INTRODUCING YOU TO RADIO

1FR-3

NATIONAL RADIO INSTITUTE ESTABLISHED 1914 WASHINGTON, D. C.



STUDY SCHEDULE NO. 1

By dividing your study into the steps given below, you can get the most out of this part of your N. R. I. Course in the shortest possible time. Check off each step when you finish it.

- □ 1. How To Study......See "How To Study" FOLDER This folder will show you how to make every minute count when studying at home, so that you can complete your N. R. I. Course in the shortest possible time. It also contains helpful instructions for answering the test questions. After reading this entire folder carefully once, follow its suggestion and read the stimulating message on the back cover of this first lesson.
- □ 2. Your Future in Radio.....Pages 1-7 of this lesson Read these pages without making any special effort to study them. Let this first reading give you a general idea of what is in the first section of the lesson.
- 4. Electricity and Magnetism......Pages 7-18 Go through this section the first time just as if it were a fascinating magazine article, without trying to study it. If this first reading makes you realize how important a knowledge of electricity and magnetism is to a radio man, you have accomplished the purpose of this step.
- 5. What Electricity ls......Pages 7-12 Re-read this section slowly once or perhaps two or three times. When you feel that you have a clear understanding of how electrons give us electricity, answer Lesson Questions 3, 4 and 5.
- □ 7. Radio Tubes.....Pages 19-26 Read this short but highly important section through the first time without thinking about studying it, just to get a general idea of what a radio tube is like. Next, read the section through slowly, mastering one paragraph at a time. Finally, answer Lesson Questions 7 and 8.
- 9. Mail Your Answers for Lesson 1FR-3 to N. R. I. for Grading.
- □10. How Radio Receivers Are Serviced......RSM Booklet No. 1 This is the first of a series of Booklets that will give you practical, how-to-do-it training in fixing radios. You will get one of these with each of your Lessons in Rad o Fundamentals. This first RSM Booklet explains the purpose of the series and shows you what is coming. Read it after you have finished Lesson 1. NOTE: Some of the Study Schedules in future Lessons may refer to "Job Sheets" instead of RSM Booklets. Read the correspondingly numbered RSM Booklet in such cases. These Booklets are an improved and enlarged version of the older Job Sheets, and are replacing them.
- □11. Start Studying the Next Lesson.

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INTRODUCING YOU TO RADIO

Your Future in Radio

DECIDING to study the N.R.I. Radio Course is one of the wisest decisions of your life. Why? Because this radio Course has been planned and written from beginning to end especially for men who must do their studying at home, usually after their

diately usable, because in the very first lessons you begin preparing for practical everyday radio jobs.

The amount of education you now have is of less importance than an unswerving ambition to succeed. The first lesson starts right at the begin-



Courtesy Westinghouse Elec. & Mfg. Co.

Courtesy Emerson Radio & Phonograph Corp.

In this very first lesson of your N.R.I. Course, you will learn the fundamental manner in which all radio tubes operate. This means that after mastering this first lesson, you will know how electrons flow through the giant high-voltage radio tubes (at the left) of station KDKA's transmitter, and you will be just as familiar with the behavior of electrons in the four tiny radio receiver tubes shown at the right.

regular day's work. You will learn radio with the aid of short-cut teaching techniques—not ordinary classroom methods, but specially-developed, thoroughly tested home-study methods. You will obtain a complete basic radio training without wasting your time in the study of unnecessary material.

You're going to get real results from this Course sooner than you expect. The fundamental radio ideas and basic testing methods which you master one after another are almost immening of radio knowledge, and prepares you for the more advanced second lesson. Each new technical or special radio word, each symbol, and each abbreviation is explained the first time it appears.

Each lesson is full of dollars-andcents information which is vital to real mastery of radio. By the time you finish the last lesson, you just can't help but be worth a whole lot more money to the world than you are today. The surest way not to fail is to determine to succeed. Make a sincere vow right now that you are going to follow up your hunch about radio that you are going to finish the N.R.I. Course and succeed in radio. Give this world-famous Course just a fifty-fifty chance, and it'll help you just as it has helped so many thousands of other men. Stick to your vow, and you won't have to worry any more about your future.

If you have faith in yourself, if you sincerely believe that you have the ability to master radio, and never waver from that belief, your battle is half won already! The N.R.I. Course will do the rest if you give it a chance to work for you. We have actual records to prove that it has brought real success to thousands of other men who had the will-power to finish it. Surely you consider yourself just as good in every respect as these men. In fact, you are probably a lot more ambitious, a lot more in earnest than some of If you are as good as these them. men, you can succeed, too.

About a hundred years ago, Abraham Lincoln said: "I will study and prepare myself, and some day I know my chance will come." These words are even more true today than they were in Lincoln's time, because there are more opportunities now and they come much faster.

In radio, for instance, there are hundreds of different job opportunities in just that one branch known as entertainment broadcasting, which deals with the sending and receiving of programs through space. You will learn about some of these jobs in the very next section of this lesson.

As an N.R.I. graduate, you will be in a position to *choose* your opportunity, instead of waiting for it to come to you. You can decide for yourself whether you want an operating job,

whether you want to do radio receiver servicing, or whether you want to play a part in building the great transmitting stations or the millions of receivers which serve this fast-growing broadcasting industry.

Furthermore, as the recently-developed frequency modulation system of broadcasting spreads throughout the country, the existing job opportunities will increase correspondingly in number. In addition to all this, commercial television is now a reality, and will be seeking trained men in increasingly greater numbers each year. Truly, the career of radio offers a promising future to men like you!

But a still more amazing fact about a radio career is that there is an even greater variety of opportunities for you outside of the entertainment broadcasting field. In short-wave radiotelephone and radiotelegraph communication systems on land, on the sea, and in the air are thousands of fascinating good-pay jobs for men who have the complete basic radio training that you are now obtaining.

In addition to all this, there are still more job opportunities for you in fields beyond the sending and receiving of messages or programs by radio.

You may some day find yourself in a theater, installing, adjusting and repairing the electric eyes and amplifier systems which make modern movies talk.

You may go to ball parks and conventions, to take charge of public address equipment which boosts man's feeble voice as much as a million times.

You may hike through canyons and foothills, using modern radio and electrical prospecting equipment to locate new oil fields and new deposits of precious minerals.

Wherever there is equipment which uses radio tubes, radio parts, radio principles and radio servicing methods, there also are opportunities for men with the thorough radio training you will soon have.

You are going to see great things happen in radio. You are going to see breath-taking new developments which far outshadow even the recent miracles of color television and noisefree broadcasting. You will be prepared for the today undreamed-of jobs which are created by new developments, because your N.R.I. training is built upon a sound foundation of fundamental radio and electrical facts.

In a way, you are to be envied, because as a graduate of this N.R.I. Course you will start your career in radio with knowledge which it has taken others almost a lifetime to acquire. Yes, one of the great things about the N.R.I. method of home training is that it gives you in a very few months even more up-to-date knowledge and information than many "old timers" have when they retire from radio!

Now you're ready to GO—ready to start your study of radio.



Radio Maps Will Guide You

At regular intervals in the N.R.I. Course, you will be given an outline of what comes next. Each outline is really a "radio map" of part of the Course, because it shows how each of the study subjects fits into your complete radio training. You will use these outlines as guides for getting your radio training as fast as possible, just as you would use a road map to guide you to a distant town over the shortest possible route.

Map of a Radio System. In this lesson, you start right in to learn about electricity, magnetism, radio tubes, and transformers—things you will work with on radio jobs every day. These subjects are so important that they deserve a radio map all their own, just for this lesson.

This first radio map takes you on a quick imaginary trip through the complete broadcasting system pictured in Fig. 1. You find out where radio tubes and transformers are used in a modern radio system, and discover that electricity and magnetism have highly important effects in each one of the six parts of the system. Also, you will begin learning about important good-pay radio jobs for which you prepare yourself.

Once you have become familiar with the practical uses for electricity, magnetism, tubes and transformers, you will be eager to tackle the rest of this lesson and learn more about each one of these subjects.

1. Microphone in Studio. Our flying trip through a radio system starts in the broadcasting studio, where the sounds which make up a radio program are produced. Sounds are mighty important to a radio man, because he has to make sure that the sounds which come out of loudspeakers in homes are as nearly as possible like the sounds which enter the microphone at the studio.

Without the aid of radio, ordinary sound waves can travel only a few hundred feet at the most. But radio changes feet into miles, and allows you to hear sounds which are hundreds or even thousands of miles away without using wires. Not many years ago this would have been called magic, but today people think nothing about hearing sounds which are halfway around the world. All this is possible simply because radio men know how to put electricity and magnetism to work!

The microphone is really the doorway of our broadcasting system. Sound waves enter it through a wire screen, and make a thin piece of metal move back and forth or *vibrate*. In the wires of this microphone are millions and millions of tiny particles called *electrons*. Magnetism gets into action and makes these electrons move back and forth in step with the vibrations of the thin metal piece.

Thus, the microphone converts sound waves into a varying electron movement. This electrical signal is called an audio signal and is actually sound in an electrical form. In other words, audio signals are electrical equivalents of sounds. (The word "audio" comes from a Latin word meaning "to hear," and is used by radio men to describe an electrical signal having characteristics of sound.)

The microphone feeds its audio signal into a *cable*, which consists of two insulated wires surrounded by braided metal which is covered with rubber.

2. Control Room. The audio signal which is produced by the microphone travels through the microphone cable to an *audio amplifier* in the control room. This audio amplifier consists of radio tubes, transformers and other parts mounted inside a cabinet and connected together in such a way that when a weak signal is fed in, the amplifier produces a new signal which is much stronger but still has the same characteristics as the weak signal.

An audio amplifier is thus a signalbooster which takes in the weak audio signal of the microphone, and puts out a new signal which is hundreds of times stronger—strong enough for the

long trip over telephone lines to the transmitter.

To be technically correct, however, we must point out that signal-boosting is accomplished by radio tubes which are capable of making more electrons move in step with the original sound. Furthermore, it is magnetism which does the work of transferring the signal from one tube to the next through transformers.

The man who adjusts the knobs at the front of the audio amplifier while watching the radio artists through a large plate-glass window and listening to their program with headphones is the monitor operator. Chatting with him for a few minutes, we learn that some announcers talk too loud for radio purposes, while others barely speak above a whisper. All speakers must sound just about the same in loudness to radio listeners in homes. so this monitor operator turns up the volume (makes the audio amplifier do more boosting) for a weak-voiced speaker, and cuts down the volume when any one starts shouting in front of the microphone.

Excessively loud sounds can overload a transmitter and cause it to go "off the air" for a few moments. To avoid this the monitor operator must turn down his volume control before a loud musical passage, a cough, or any other loud sound. Some operators become so skilled at figuring out when loud sounds will occur that they are jokingly called mind-readers!

3. Transmitter. Leaving the control room, we follow our radio program (now in the form of a strengthened audio signal) over a telephone line to the transmitter building which is usually located a few miles outside the city.

Once inside the transmitter, we find that our audio signal gets another



FIG. 1. This first radio map shows the six important sections of a radio broadcasting system. Start with the studio microphone at 1 and trace the signals until you reach the receiver loudspeaker at 6.

tremendous boost in strength from tubes and transformers. But no matter how much we boosted this audio signal, we wouldn't be able to make it travel more than a few hundred feet through space at the most.

We have to carry our audio signal through space on another signal known as the *carrier signal*. This carrier signal is the true *radio* signal.

Each station is assigned a carrier signal of its own by the Government. The carrier signal is produced by more radio tube circuits, and is combined with the audio signal in a radio tube circuit designed especially for "mixing" purposes.

The audio signal, riding on its own

radio carrier, is given a final boost in strength by huge power tubes, and is then fed to the transmission line which serves the transmitting antenna. There's plenty of power behind our signal now—enough to fry to a crisp any unlucky mouse or grasshopper who gets across the high-voltage tube terminals.

The radio operator on duty at the transmitter must see to it that the radio signal which the transmitter is sending out into space has the same form at all times as the original sounds in the studio.

In addition, the operator must check meter readings in the transmitter regularly and try to anticipate possible breakdowns of tubes and parts. He can then replace weakened parts after the station shuts down for the day.

The goal of an operator is to keep his station on the air every second of the scheduled operating time. This is not as difficult a job as it seems, for an operator who fully understands the simple fundamental characteristics of each radio part and circuit in the transmitter rapidly acquires an uncanny ability to tell when a particular part is failing and in need of attention or replacement.

4. Transmitting Antenna. Coming out of the transmitter building, we hike along the transmission line to a little house just under the transmitting antenna tower. Technically speaking, we should call this the tuning house, but since you plan to be a practical radio man, you might as well get accustomed to hearing it called the "dog house." Every transmitter has its dog house in the shadow of the antenna tower, and nearly every radio operator has at one time wisecracked about "leading a dog's life" when he had to go out to the dog house in stormy weather to read the meters.

The dog house contains radio parts which can be adjusted so as to "tune" the tower to the assigned carrier signal of the station. The antenna then gets all the signal power which the transmitter can feed to it through the transmission line. The meters in the dog house tell "how the station's doin'," and must be read at regular intervals each day.

From the tuning house, our radio signal goes right up the giant antenna tower, to the very top. The signal still consists of electrons moving back and forth, back and forth, the entire length of the tower, and these electron movements through a tower in the sky produce the radio waves which travel off into space away from the tower.

Our radio program is now being carried out through space in all directions by radio waves.

You will learn in the next lesson that a radio wave is just a combination of electric and magnetic effects, varying in strength from instant to instant just as the pressure of the original sound wave at the studio is varying.

5. Receiving Antenna. Through the sky we trace the radio waves to the receiving antenna at some far-off home. As these waves move past the receiving antenna wire, the millions of electrons which are always in a metal wire start moving back and forth in exactly the same way as did the electrons in the transmitting antenna tower. The receiving antenna thus changes our radio waves back into an audio signal riding on a carrier signal. It is like the signal we had at the output of the transmitter, except that the signal is millions of times weaker due to its trip through space.

6. Receiver in Home. We follow our signal down the antenna lead-in wire, and we arrive at the radio receiver in the home.

We find that the first few sections of the receiver build up the strength of this weak signal, step by step. The carrier signal, no longer needed, is next removed from the audio signal by a section called the *detector*. After this, the audio signal is boosted in strength by more radio tube circuits in the *audio amplifier* section. This section feeds our signal into the very last transformer in the receiver, called the *output transformer*.

Finally, we come to the all-important loudspeaker, which is connected to the output transformer. The loudspeaker uses both electricity and magnetism to change the greatly strengthened audio signal back into sounds. Radio servicemen are highly important in connection with this last section of our broadcasting system, for these men are needed to keep the millions of receivers in this country in operating condition. They have to locate and replace defective parts, and make correct adjustments of receivers whenever necessary, so the sounds coming out of the loudspeakers will be as nearly as possible like the original sounds in the studio.

Electricity and Magnetism

Even in a brief study of a radio system, we cannot help but marvel at the important parts which electricity and magnetism play in the operation of practically every radio device. But what is electricity? What is magnetism? If you learn the answers to these two questions once and for all now, you will have the foundation for everything else you will study and work with in your radio career.

Our first question is about electricity. But don't expect to find a long and complicated explanation of electricity—don't even expect to find the rest of this lesson devoted to this one subject. This is a *practical* home study Course, so it gives you only those electrical facts which practical radio men need to know. Once you master these few facts about electricity—once you realize that all forms of electricity are due either to electrons in motion or electrons at rest—you'll be ready to make electricity work for you in radio equipment.

What Electricity Is

Everything on this earth is made up from tiny particles, so small they cannot be seen even with the most powerful microscope. These particles are attracted to each other with great force in the case of solid materials like iron or copper, with less force in the case of liquids like water or oil, and with the least force in the case of gases like oxygen or hydrogen. Scientists have identified several different kinds of these particles, but the only one in which radio men are interested is the electron. It has been proved beyond all doubt that electrons exist everywhere—in the air, in water, in every single object in the world!

During normal conditions, each radio part, each wire, each object on the earth has its own definite number of electrons. The exact number runs way up into the billions even for small objects, but always depends upon the size and nature of the object. Whenever this normal condition exists for some object, that object has no electricity, and is said to be *neutral* or *uncharged*.

In metals like copper, silver or aluminum, some of the electrons are continually wandering around inside the metal itself. These electrons are known as *free electrons*, because they are free to move. The rest of the electrons always stay in the same locations; in fact, if we could discover how to move them, we would be able to change one metal to another—we would realize the age-old dream of changing lead into gold.

We can secure electricity either by taking some of these free electrons away from an object or by adding more free electrons to the object, because we then upset the normal balanced condition. This brings us to an important basic definition:

Whenever an object has more or fewer free electrons than normal, electriciter exists in the object.

Sources of Electricity. But where can we get extra free electrons? This is an easy question to answer-get the extra free electrons from a battery, a power line, a generator, or any other device which possesses the ability to supply free electrons. Examples of a few common sources of electricity are pictured in Fig. 2.

Since radio men always work with free electrons (never with the fixed electrons), we can just say electrons from now on, when we mean free electrons. Remember-whenever you read about electrons in radio magazines or N.R.I. lessons, think of free electrons.



FIG. 2. Examples of sources of electricity used by radio men to make electrons move in radio circuits. Storage batteries are used with radio receivers in automobiles, ships, airplanes, and farm homes. Radio batteries are used chiefly in portable receivers. Power line outlets serve by far the greatest portion of the radio receivers in this country. Flashlight cells are used in small pocket-size portable receivers. Wind-driven generators keep radio storage

batteries charged in farm homes.

Law of Electric Charges. Technically speaking, a law is simply a statement which always holds true. There are only a few of these laws which are important to a practical radio man. Once you understand these laws, you can use them to figure out how radio parts and circuits will behave under various conditions.

The first law which we take up is one which tells how electrons will behave when near other electrons or when near electric charges. An electric charge is simply a quantity of electricity.

There are two kinds of electric charges, quite unlike, and for convenience they are called positive charges and negative charges.

The electron is a negative charge. and is also the smallest amount of electricity which can exist by itself.

An object or terminal has a *negative* charge whenever it has more electrons than normal. An object or terminal has a *positive charge* whenever it has fewer electrons than normal.

Two electric charges cannot exist side by side without trying to move each other. Nature has its own law which tells which way they will move. If the two charges are alike (if both are negative or if both are positive), they will *repel* each other, or tend to push each other away. If the two charges are unlike (one is negative and the other is positive), they will attract each other and tend to move together.

Let's repeat this law now in threeword statements which will be easy for you to remember:

LAW OF ELECTRIC CHARGES Like charges repel. Unlike charges attract.

Once you learn this simple law, you will have the true basic idea of

how all radio tubes work. You will find that this law helps you to understand the use of test instruments, and aids you in many other ways when you are working on a radio receiver, transmitter or other piece of radio apparatus.

Here is a helpful suggestion: Write this law on a scrap of paper now, and place this paper where you will see it once in a while during the next few days.

The battery terminal which has fewer electrons than normal is called the positive terminal of the battery. On diagrams and on batteries, we use a plus sign (+) to indicate that this terminal is positive.

Nature always tries to restore things to normal. That's the reason why the electrons on the negative terminal want to get over to the positive terminal of the battery. Yes, there's just one place those electrons



Sixty billion electrons A free electron can letter"i"shown above. sponge full of holes.

side by side would be travel through metal needed to form a line as easily as an ant across the dot in the can travel through a

Battery Terminals. An ordinary battery, such as a dry cell or an automobile storage battery, always has two terminals, one having more electrons than normal and the other having less electrons than normal. (Electrons. remember, are negatively charged particles.) The battery itself produces this condition by a chemical action which need not be studied here.

The battery terminal which has electrons (more negatively more charged particles) than normal is called the *negative terminal* of the battery. On radio diagrams, as well as on actual batteries, we use a minus sign (-) to indicate that this terminal is negative.

as rocks when you're standing a hundred feet away from them, and

Sponges look as solid A piece of metal would look as full of holes as a sponge if our eyes could see its actual metal structure.

on the negative terminal want to goto the positive terminal of their own batteru.

When we connect a wire between the two terminals of a battery, we secure a complete circuit. Here is what happens. The instant the circuit is completed by touching the wire to the — terminal, the extra electrons on this terminal tend to rush into the wire because they "sense" a clear path ahead to the + terminal. These fastmoving electrons from the battery bump into free electrons in the wire, and push the free electrons forward toward the positive terminal. In no time at all, every electron gets bumped. All along the wire, electrons begin pushing other electrons toward the + terminal. Each time an electron is pushed into the + terminal, another electron enters the wire from the -terminal.

Suppose that we connect a length of wire to the positive terminal of an ordinary storage battery, as shown in Fig. 3A. Our wire contains great numbers of free electrons which are ready to "go places," but nothing can happen yet. Our wire is merely an extension of the + terminal.

Suppose, however, that we connect the other end of the wire to the negative terminal of the storage battery, as shown in Fig. 3B. We get action immediately. The wire gets warmer and warmer, a sure sign that electrons are moving.

Of course, a radio man would warningly point out that we are "shorting" the battery with the wire, but we can do this for a *short* time without seriously draining a good storage battery.

Which way are the electrons moving through our wire? Well, if the negative terminal wants to get rid of electrons because it has too many, and the positive terminal wants more electrons to make up its shortage, you should be able to figure out the answer yourself.

Yes, electrons move from the negative terminal to the positive terminal through the wire, just as is shown by the arrows in Fig. 3B. This agrees with the law of electric charges, which says that electrons will be attracted by a positively charged terminal. Thus, we've made use of this law already!

The movement of electrons through a complete electrical circuit can be compared to the movement of marbles through the length of tubing shown in Fig. 4. Each time you push a marble into one end of a filled tube, an entirely different marble pops out of the tube at the other end.

As soon as electrons move from the — terminal into the wire, the battery forces more electrons to the — termi-



FIG. 3. Nothing happens when a wire is connected to only one terminal of a battery. There must be a complete path for electrons between the two battery terminals in order for electrons to flow. CAUTION: Never "short" a battery needlessly with a wire.

nal. This keeps up as long as we leave the circuit complete or until we have used up all the power stored in the battery. It is because a battery can supply electrons in this way that we consider batteries as sources of electricity.

The movement of electrons through a complete circuit is called an *electric current* or simply a *current*. When the source of electricity forces electrons through a circuit in the same direction at all times, the resulting flow of electrons is known as a *direct current*, abbreviated as *d.c.* The more electrons we have moving past a given point in the wire each second, the greater is the value of direct current flowing through the wire.

Speed of Electricity. Whenever a wire is connected to the two terminals of a battery so as to give a complete circuit, the free electrons in the wire begin moving toward the positive terminal. These free electrons get into motion almost instantly, hence we say that the effect of electricity travels almost instantly from one point to another. (The actual speed of this electrical effect has been measured by scientists and found to be about 186,-1000 miles per second, which is the same as the speed of light.) The first effect of electricity travels at this speed regardless of battery strength.

Although the effect of electricity travels so rapidly during the first instant when a connection is made, the free electrons move slowly through the wire (only a fraction of an inch per second) once they are in motion. It is just as if we had a long line of soldiers, each standing with his bayonet pressed against the man ahead. When the man at the rear of the line moves forward, the entire line must jump into motion almost instantly so no one will be stabbed in the back. Once in motion, however, the soldiers march at a slow and steady speed corresponding to the slow movement of electrons through a wire.

The instant we complete the circuit, the free electrons in the wire start moving, and continue moving at the slow, steady rate determined by the strength of the battery. If we increase the strength of our battery more electrons will-move out of the negative terminal each second and hence the

current will increase. If we reduce the strength of the battery, the current will decrease. In radio, we often adjust the strength of each battery so exactly the desired number of electrons will be moving through each part in the circuit per second.

There is a need for a convenient way in which to specify the strength of a battery or other source of electricity. Radio men use the term volts for this purpose. Thus, you will often hear radio men speak of a 6-volt storage battery, a 115-volt power line,



FIG. 4. When you push marble A into one end of this marble-filled tubing, marble B will pop out of the other end. Although marble B will move instantly, it will take some time before enough marbles can be pushed in to make marble A come out at the other end. Continuous electron flow through a wire occurs in much the same way.

a 45-volt B battery or a $1\frac{1}{2}$ -volt A battery. In later lessons you'll learn a lot more about what volts mean in connection with these voltage sources. For the time being, just remember that volts tell you how strong a source of electricity is.

A voltage which sends a direct current through a circuit is known as a *direct current voltage*, usually abbreviated as d.c. voltage. All batteries produce d.c. voltages. An automobile generator is also a d.c. voltage source.

Polarity. When giving the strength of a d.c. voltage source, it is often

desirable to tell which terminal is positive or which is negative. We are then telling what the *polarity* (pronounced po-LAIR-i-tee) of the voltage source is. Thus, when a radio man says, "this wire of your auto radio should be connected to the + terminal of the car battery," he is telling what the *polarity* of the connections to the voltage source should be.

Conductors. The wire used for connecting parts together in radio circuits is made from copper. The wire used in making radio coils, transformers, loudspeakers and many other parts is likewise copper. The reason this metal is used so much in radio equipment is that it has large numbers of free electrons.

Silver, aluminum and brass are other metals which have plenty of free electrons. These metals and copper are known as *conductors*, because they pass or conduct electricity. A good conductor is said to have *low resistance*, because it offers little opposition or *resistance* to the flow of electrons.

Insulators. To make electrons work for us, we must make correct paths for them by using conductors like copper wire, and we must prevent these paths from touching undesirable paths. In other words, we must separate or *insulate* our desired wire paths from all other paths, so that electrons will go through every part in the path instead of taking short-cuts through other parts.

A wire can be insulated by covering it with an *insulator*, a material which offers a great deal of opposition to movements of electrons.

Silk, cotton, enamel, rubber, paper and glass are good insulators, because they are materials which have very . few free electrons.

Free electrons have just as difficult a time in moving through an insulator as soldiers would have in moving through a barbed wire entanglement. Just as barbed wire is used to limit the movement of soldiers, so are insulators needed in radio circuits to make electrons take the correct paths through radio parts and wires. In other words, insulators offer high resistance to the flow of electrons.

An insulator can become a conductor if the voltage is made high enough to cause actual destruction of the material. This breakdown is accompanied by an electric arc or spark which burns the insulating material, thereby providing a conducting *path* for electrons.

This is exactly what happens when the insulation in a radio part breaks down; the radio man simply says the part is *shorted*, but actually the insulation fails due to the heat of the spark, and electrons flow through the resulting charred hole in the insulation. The action is much like that of a tank traveling through a barbed wire entanglement; once the tank has forced its way through, soldiers have no difficulty in following.

Radio Uses for Magnetism

Without magnetism, there would be no radio industry at all, for magnetism and electricity working together give us radio waves to carry messages, music and programs through space.

There are also quite a few radio parts which could not work without magnetism. Both magnetic loudspeakers and modern permanent magnet dynamic loudspeakers use powerful permanent magnets to help make the large paper cone move and produce sound. Each headphone unit uses a tiny but powerful permanent magnet to help make the round metal disc move and produce sound.



An electrodynamic loudspeaker has an electromagnet, a device which acts as a magnet only when current is actually flowing through the magnet coil.

Electromagnets are used in some television picture tubes to make a beam of electrons sweep back and forth in the correct manner for "painting" a picture or pattern on a screen.

You will also find electromagnets in auto radio vibrators, in the relays which do automatic switching work in radio transmitters, and in many special radio parts, which are taken up in later lessons.

Radio transformers depend upon magnetism for the transfer of signals and electric power from one circuit to another. The silent magnetic action in a transformer is just as important in a radio system as the humming sound of a vibrator, the chatter of a loudspeaker, or the clicking of a relay.

Since magnetism is so important to a radio man, let us pry into its secrets now, to see what it really is and how it can work in so many different ways in radio parts.

Permanent Magnets. The ancient Greeks discovered, near the city of Magnesia in what is now Turkey, a mineral which possessed the unusual ability to attract small pieces of iron. The Greeks called this material magnes, and from this term comes the word magnet which we know today. Early experimenters found that they could make needles and other pieces of hard steel act like magnets simply by rubbing these steel objects on a piece of this magnetic ore. These magnetized objects were called permanent magnets, because the magnetic characteristics seemed to be permanent.

When a magnetized steel needle is placed on a cork which is floating in water so it can turn easily, the needle will always line up in a direction corresponding closely to north and south. This magnetic experiment is illustrated in Fig. 5. It led to the first practical use for magnets, in the compasses of



FIG. 5. If a magnetized steel needle is placed on a floating flat piece of cork, the needle will line up in the north and south directions like a compass, because the needle lines up with the magnetic field of the earth. In radio, we produce magnetic fields much stronger than those of the earth, and use these fields to make iron or steel objects move or line up in desired ways. early sea voyagers and travelers. For practical reasons, the needle in a compass is mounted on a delicately pivoted bearing which turns just as easily as the cork in water.

Poles of a Magnet. The ends of a permanent magnet are called *poles*, probably because these ends point toward the *poles* of the earth when a magnet is pivoted. The magnet pole which points north is called the *north*

net is brought near the S pole of the compass needle. Our conclusion is that magnet poles which are alike repel each other.

If the S pole of a bar magnet is brought near the N pole of a compass, a strong attracting action is noticed, just as is indicated in Fig. 6B. Thus, unlike magnet poles attract each other.

This law of magnetism is fixed by nature, and always holds true. It is



FIG. 6. A simple method of demonstrating the law of magnetism. This law applies to the operation of loudspeakers, headphones, relays, vibrators, motors and many other radio parts.

pole of the magnet. The magnet pole which points south is called the *south* pole. On diagrams of magnets, the north pole is usually marked N, and the south pole is usually marked S.

One law which is particularly useful to a radio man is the law which tells how magnets will behave. It can be demonstrated with a compass needle and a bar magnet (a magnetized bar of steel) in the following manner:

If the north (N) pole of the bar magnet is held near the north pole of the compass, just as is shown in Fig. 6A, a repelling action is noticed because the compass is itself a small bar magnet. This same repelling action occurs when the S pole of the bar magrepeated here in two simple statements which will be easy for you to remember:

LAW OF MAGNETISM

Like magnetic poles repel. Unlike magnetic poles attract.

Again you may want to write this law on a piece of paper so you can look at it often.

Magnetic Lines of Force. If a thin sheet of cardboard is placed over a bar magnet, and iron filings are sprinkled evenly over the cardboard, the iron filings will arrange themselves in definite lines, as shown in Fig. 7A. If the experiment is repeated when the north poles of two bar magnets are end

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to end, the pattern shown in Fig. 7B is obtained. If the experiment is repeated once more, with opposite poles of the two bar magnets near each other, the pattern shown in Fig. 7C is obtained.

These experiments show definitely that there is some kind of force acting through the space surrounding a magnet. The scientist Faraday studied magnets for a long time, and finally decided that these forces must act along invisible but definite lines or paths in space. He said that these paths were the same as those traced out by the iron filings, and gave the name "magnetic lines of force" to the natural forces which a magnetic pole produces in space. Radio men still use this name today, and also say that all the magnetic lines of force around a given magnet make up the magnetic field of the magnet.



FIG. 7. Here are some of the results which can be obtained by placing sheets of cardboard over permanent bar magnets, then sprinkling iron filings over the sheets. Magnetic lines of force are traced out by the filings almost magically.

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FIG. 8. A small compass can be used to trace magnetic lines of force near a permanent magnet.

When a small compass (which always has an N pole and an S pole) is moved to various positions in the vicinity of a bar magnet, as indicated in Fig. 8, the compass needle will line up with the magnetic lines of force at each position, and will thus trace out the magnetic lines of force in the vicinity of the bar magnet. Compare Fig. 8 with Fig. 7A, and you will see that the compass and iron filing experiments check perfectly.

Magnetic lines of force are considered to have a *direction* as well as a position in space. This direction is always that in which the N pole of a compass will point. Thus, the arrows in Fig. 8 show that magnetic lines of force come out of the N pole of a magnet, and enter the S pole. This rule is worth repeating here for reference purposes, even though you do not have to memorize it.

Magnetic lines of force come out of the N pole of a magnet, and go into the S pole. The N pole of a compass (painted a dark color) always points in the direction of magnetic lines of force.

Electromagnets. In 1820, the scientist Oersted discovered that when an electric current is flowing through a wire, the current produces magnetic effects in the space around the wire. When he moved a compass around a current-carrying wire, the compass needle definitely traced magnetic rings (circular magnetic lines of force) as shown in Fig. 9A. (Only two rings are shown in Fig. 9A, but actually there are many more magnetic rings surrounding a current-carrying wire, and these rings are of many different sizes.)

This experiment with a compass and a current-carrying wire proves that electrons in motion produce magnetic effects similar to those of a bar magnet.

The direction of the magnetic lines of force around a current-carrying wire depends upon the direction in which electrons are moving through the wire, and can be determined almost instantly with the Left-Hand Rule given here and illustrated in Fig. 9B. This rule is well worth remembering.

LEFT-HAND RULE FOR A CURRENT-CARRYING WIRE

Imagine you are grasping the wire with your left hand in such a way that your thumb points in the direction of electron flow. Your fingers, curled around the wire, will then be pointing in the direction of the magnetic lines of force.

The magnetic lines of force which surround just a single current-carrying





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FIG. 10. When we send electrons through a loop of wire, the magnetic lines of force all go through the center of the wire loop in the same direction.

wire are weak and do not have much practical use, but we can concentrate these lines of force and make them more useful by bending the wire into a loop or single-turn coil, as shown in Fig. 10. Now all the circular magnetic rings pass through the center of the coil in the same direction and reinforce each other.

If we add more turns of wire to this coil, as in Fig. 11, the magnetic field inside the coil becomes much stronger even though the same current is still flowing through the wire. For this reason, radio coils usually have a great many turns of wire.

Whenever a coil like that shown in Fig. 11 is carrying a direct current, this coil acts exactly like a permanent magnet. If we trace out the magnetic lines of force with a small compass, we find that they all go into the righthand end of the coil and come out of the left-hand end. The right-hand end of the coil corresponds to the Spole of the permanent magnet shown in Fig. 8, so we call this end the S pole of the coil. Similarly, we call the lefthand end of the coil the N pole.

The more current we send through a given coil, the greater will be the

strength of its magnetic field. This is logical, for more current means more electrons moving through the wire, and each additional electron in motion adds its share of magnetic rings to those passing through the center of the coil.

When a piece of iron or soft steel is placed in the center of a currentcarrying coil, the magnetic field which passes through the coil becomes much stronger, because soft steel is a much better path for magnetic lines of force than air. Since the magnetic field exists only when electric current is flowing, coils with iron cores like this are called *electromagnets*.

The magnetic effect of an electromagnet is only temporary, for it disappears almost as soon as the current flow through the coil is stopped. This characteristic makes electromagnets more desirable than permanent magnets for many radio parts.

Any one who wants to demonstrate how an electromagnet picks up iron objects like tacks can make one from ordinary materials. Simply wind about 25 turns of insulated copper wire around a large nail, then connect the ends of the wire to the two terminals





of a flashlight cell and bring the nail near the tacks. (Leaving this electromagnet connected longer than a minute or two will drain the battery.)

The strength of an electromagnet having a particular size of iron core depends chiefly upon two things, the number of turns of wire used in the coil and the amount of current flowing through the coil. Increasing either one will make the electromagnet stronger. Reducing the current or



FIG. 12. Magnetic story of a bar of iron. The small arrows represent invisibly tiny magnets inside the bar.

removing some of the turns of wire will make the electromagnet weaker.

The magnet in every electrodynamic loudspeaker is an electromagnet. The coil of this electromagnet is known as the *field coil* of the loudspeaker, because it produces the *magnetic field* required to operate the loudspeaker. An electrodynamic loudspeaker will work only when current is flowing through the field coil, so this type of loudspeaker will always have two extra wires for this field current.

Explanation of Magnetism. What makes it possible for metals like iron to become magnets? The answer which scientists have for this question is that each invisible tiny particle of iron is a tiny permanent magnet. Ordinarily these tiny magnets are arranged "helter-skelter" in iron, as indicated in Fig. 12A, so that their magnetic fields act in all directions and cancel each other. • If we place an iron bar in a strong magnetic field such as that existing inside a current-carrying coil, the tiny magnets within the bar will tend to line up with the magnetic lines of force of the coil. When this occurs, the entire bar becomes one large and powerful magnet, and we have the electromagnet illustrated in Fig. 12B. When the magnetizing force is removed by stopping the current or taking the iron bar out of the coil, practically all of





the magnets return to their original helter-skelter arrangement.

If a bar of hard steel is used instead of iron, however, the magnets remain in their new lined-up position even after the magnetizing force is removed, and we have a permanent magnet. The more tiny magnets we can line up in one direction, the stronger is the permanent magnet.

Iron is the most useful magnetic metal. Nickel and cobalt (a silverwhite metal resembling nickel) have weak magnetic characteristics by themselves, but when one or both of these pure metals are mixed with other metals in definite ways to give magnetic alloys, the results are even better than are possible with steel. One example is *alnico*, a combination of *aluminum*, *nickel* and *cobalt* which is now widely used for permanent magnets in loudspeakers. Practically every piece of radio equipment being made today contains at least one radio tube, so there should be no question whatsoever regarding the importance of learning about tubes as soon as possible in your course. You already know where tubes are used in broadcasting systems—you know the vital facts about the electrons which make radio tubes possible —so now you're fully prepared to learn what radio tubes are like and how they work.

First Amplifier Tube. The invention by Lee DeForest in 1904 of the "audion" radio tube shown in Fig. 13 did more than anything else to make possible radio as we know it today. With tubes like this, it was possible for the first time to *amplify* a weak radio signal and thereby build up its strength hundreds or even thousands of times.

This first radio amplifier tube is remarkably simple, as you can see for yourself by examining the sketch in Fig. 13. There are only three main parts or *electrodes* (pronounced *ee-LEK-troads*). Each electrode has its own name.

The *filament* of this tube is just like the filament of an ordinary electric lamp. When current is sent through this filament, it gets hot. On the surface of this filament is a chemical coating which actually gives off or *emits* electrons when heated. A filament which supplies free electrons in this manner is also serving as the *cathode* electrode.

The rectangular plate of metal in this tube is called the *plate*. When a voltage source is connected between the plate and the cathode in such a way that the positive terminal goes to the plate, the plate attracts the electrons which are emitted by the cathode. Under this condition, the plate electrode is said to be positive with respect to the cathode.

The grid is the important electrode introduced by DeForest; it is between the filament and the plate, in a posi-





tion where it can be made to have a powerful control over the number of electrons which get to the plate.

Many additional parts have been added to this first DeForest "audion" tube and many improvements have been made, but modern radio tubes still use the same control grid idea.

Modern Tube Construction. A cut-away diagram showing the con-

struction of a typical modern radio tube is shown in Fig. 14.

In the center of the tube is the *filament*, which becomes -red-hot and supplies heat.

Surrounding the filament is a cylindrical or tubular-shaped electrode called the *cathode*, which emits electrons when heated by the filament.

The remaining two electrodes encircle the cathode. The outermost is a metal cylinder which attracts the electrons and is still called the plate even though it has no resemblance to the flat metal plates of early radio tubes. Between the cathode and the plate is a spiral wire arrangement called the grid, which is used to control the number of electrons moving from the cathode to the plate.

The electrodes of a vacuum tube are enclosed either in a metal or glass housing called the *envelope*, from which all air has been pumped out during manufacture so as to give a vacuum, necessary for maximum electron emission.

The tube electrodes are connected to prongs on the tube base. This base fits a socket which provides individual contacts for each electrode.

The filament is considered an electrode of a radio tube only when, in addition to its job of providing heat, it has the extra job of giving off electrons. Modern 1.4-volt battery-operated tubes and the old DeForest audion are examples in which the filament also serves as the cathode.

The least number of electrodes a tube can have is two: 1. The cathode, which emits (gives off) the electrons; 2. The plate, which attracts the emitted electrons. There is no definite limit to the maximum number of electrodes, but radio receiving tubes usually have somewhere between two and six electrodes. A tube can have two or more different types of grids, but for the present we will consider only tubes having one grid.

The tube in Fig. 14, which we are using as an example, has three electrodes—a cathode, a plate, and a grid. For this reason, it is known as a triode (pronounced "TRY-oad"). Let us examine the important parts of this vacuum tube individually now, and see what voltage each part requires.

Filament. The filament in a tube like that in Fig. 14 is a fine hairpinshaped wire, with heavier wire leads going from the ends of the filament to the two filament prongs on the tube base. The fine filament wire becomes red-hot when filament current is sent through it by a battery or other voltage source. This filament merely provides heat, so it cannot be called a cathode.

The filament voltage source is commonly designated by the letter "A." Thus, either an A battery or some other A voltage supply can be used to send current through the filament. In those battery receivers where the polarity of the filament voltage is important, the positive terminal is identified as "A+," and the negative terminal is marked "A-."

Cathode. Surrounding the filament of our tube is a metal sleeve about $\frac{1}{16}$ inch in diameter, serving as the *cathode*. This cathode sleeve is coated with special chemicals, which have the ability to give off large quantities of free electrons when they are heated by the red-hot filament. Remember that the cathode is the electrode which emits electrons.

Plate. The metal cylinder farthest away from the cathode is the plate, sometimes known as the anode. The plate in a vacuum tube attracts the electrons which are emitted by the cathode, because the plate voltage source makes the plate highly positive with respect to the cathode. This is a practical example of the Law of Electric Charges; the electrons are negative and the plate is positive, so we have *unlike* charges which *attract* each other. The plate is firmly mounted and can't move toward the electrons, so the electrons move toward the plate.

Plate Current. When a vacuum tube is properly connected to its voltage sources, the electrons emitted by the cathode will be attracted by the plate. This electron flow (current) through the tube from cathode to plate is commonly called the *plate current* of the tube. The more positive the plate voltage is with respect to the cathode, the *higher* will be the plate current.

The plate voltage source is commonly designated by the letter "B". Thus, either a B battery or some other B voltage supply must be used to provide the required high positive voltage for the plate. The positive terminal of the B voltage supply is connected to the plate, and the negative terminal is connected to the cathode.

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Radio men use the notation "B+" to identify the positive terminal of the plate voltage source and the plate *lead* to which this terminal should be connected. (A *lead*, pronounced "*leed*," is a wire used for connecting purposes.) Likewise, "B-" is used for the negative terminal of the plate voltage source and the cathode lead to which this negative terminal should be connected.

The Grid. A radio tube must have one additional electrode besides the plate and the cathode before it can act as an *amplifier* for electrical signals. This additional electrode is the grid; it is usually in the form of a coil of fine wire mounted between the cathode and the plate, with considerable space between the turns of wire, as shown in Fig. 14. The name "grid" comes from the fact that in early radio tubes, this in-between electrode was a grid-shaped wire (see Fig. 13).

When there is no electrical charge (no voltage) on the grid, electrons can readily pass between the grid wires when traveling from the cathode to the plate.

When a charge is placed upon the



Courtery National Union Radio Corp.

FIG. 14. Cut-away sketch showing the construction of a typical radio tube. Part of the cylindrical plate electrode is cut away to show the grid and cathode, and part of the cathode is cut away to show the filament wires.

grid of a vacuum tube, however, the grid controls the number of electrons which move from the cathode to the plate. This charge on the grid is commonly called the *grid bias voltage*.

If a positive charge is placed upon the grid, the grid will attract electrons and speed them up, so that more electrons pass through the grid wires and reach the plate. If a negative charge

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is placed on the grid, it will repel electrons and reduce the number which reach the plate. This is simply an application of the fundamental electrical law that *like charges repel and unlike charges attract*.

In radio work, the grid usually is made *negative* with respect to the cathode, because a negative grid bias voltage permits more nearly perfect amplification of signals.

When a negative charge is placed on the grid, so it is more negative than the cathode, the electrons on the grid repel the free electrons coming from the cathode, and actually prevent some of these free electrons from passing through the grid wires. The greater the negative charge on the grid of a radio tube, the less is the electron flow through the tube.

We can place a negative charge on the grid of a vacuum tube simply by connecting the negative terminal of a voltage source to the grid, and connecting the positive terminal of the voltage source to the cathode. The grid is then *negative* with respect to the cathode.

The grid voltage source is commonly designated by the letter "C," with the terminals being marked "C+" and "C—" respectively. Thus, a C battery can be used to make the grid negative with respect to the cathode. Radio men say then that the grid has a *negative grid bias voltage* or a *negative C bias voltage*.

Operating Voltages. We have thus seen that a modern three-electrode (triode) radio tube requires certain definite voltages. These are known as operating voltages, and can be summarized as follows:

1. The *filament voltage*, required to send current through the filament for heating purposes.

2. The plate voltage, required to make the plate highly positive with

respect to the cathode, so the plate will attract the electrons emitted by the cathode.

3. The grid bias voltage, required to make the grid negative with respect to the cathode, so as to permit more nearly perfect amplification of signals.

More Tube Facts. We often deal with very weak signals, which must be boosted in strength (in voltage) without changing their nature. A radio tube is one device which can do this job successfully (a transformer is another).

With a definite grid bias voltage, a definite plate current flows through the tube. When we apply a weak signal to the grid of a radio tube along with the grid bias voltage, this weak signal (still varying in strength from instant to instant) makes the grid voltage change rapidly above and below its definite grid bias value. As a result, the plate current of the tube varies above and below its original value in exactly the same way that the weak signal is varying. This varying plate current gives us an entirely new signal, much stronger but still having the same nature as the weak signal. This is how it is possible for a radio tube to amplify weak signals.

In Great Britain, radio tubes are called *valves*. This is quite an appropriate name, because the grid does act as a sort of valve when it increases or decreases the stream of electrons flowing from the cathode to the plate.

Now what is the practical use for all this information about radio tubes? Here's the answer—it prepares you to understand the next lesson, which goes right ahead to give you a great deal of interesting and practical information about radio tubes at work.

Vacuum Tube Symbols. It is inconvenient to draw a picture or sketch



FIG. 15. Symbol for a two-electrode radio vacuum tube, as you might see it in textbooks, magazines and service manuals (A), and as you and other radio men might draw it free-hand for practical use (B).

of a vacuum tube each time we wish to show it in a circuit diagram. For this reason, radio men use simple designs or symbols to represent vacuum tubes as well as other radio parts. Symbols are a "shorthand" system of showing radio parts in simple form.

The symbol for a two-electrode vacuum tube having a heater-type cathode (a cathode heated by a filament) is shown in Fig. 15A. The filament is shown as a V-shaped wire projecting inside the circle which represents the glass or metal envelope of the tube. The cathode is shown as a heavy line tending to surround the filament, just as it does in an actual tube. The plate is another heavy, straight line, parallel to the cathode and some distance away from it. Thus, you can see that each line on a symbol tells its own story.

Although tube symbols are carefully drawn with compass and ruler for diagrams which appear in textbooks and magazines, the radio man can make equally understandable symbols freehand. For example, if a practical radio man were to draw the tube symbol of Fig. 15A while explaining some circuit to a radio-minded friend, the symbol would probably look like that shown in Fig. 15B. Try sketching this symbol roughly yourself a few times.

The symbol for a three-electrode vacuum tube having a heater-type cathode is shown in Fig. 16A. The grid is shown on this symbol as a dashdash line located between the cathode and the plate. When drawn freehand, this symbol will appear somewhat as in Fig. 16B. Try sketching this roughly a few times, pronouncing to yourself the names *filament*, cathode, grid and plate as you draw them in.

Battery Connections. A threeelectrode radio tube requires three separate voltages. When these are provided by batteries as they are in modern portable receivers, we need an A battery for the filament, a B battery for the plate, and usually a separate C battery for the grid. These batteries all deliver the same kind of electricity (d.c. voltage), but the batteries vary in size because different voltages are required and because some parts of the tube draw more current than others.

The actual battery connections for a radio tube might be as shown in Fig. 17A. The triode tube is shown in symbol form because a picture of a tube would not tell you how the connections were made to the electrodes.



FIG. 16. Symbol for a three-electrode radio tube. Practice drawing it free-hand. as at B.

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FIG. 17. These three types of diagrams each contain exactly the same amount of useful information about a vacuum tube circuit. Obviously, the diagram at C is the simplest to draw; for this reason, you will encounter diagrams like it most often in your radio studies and radio work. There is no need to make fancy diagrams like that at A when a simple diagram will serve just as well.

As you can see, the A battery in Fig. 17A is connected to the two filament leads. The B battery is connected between the plate and the cathode, so as to make the plate positive with respect to the cathode. The C battery is connected between the grid and the cathode, so as to make the grid *negative* with respect to the cathode. Thus, both the + terminal of the C battery and the - terminal of the B battery are connected to the cathode. Each battery acts only in its own circuit, however, because the electrons which leave the -- terminal of a battery always go to the + terminal of that same battery.

There is no need to draw a pictorial sketch of each battery, however. We can tell just as much with the battery symbols shown in Fig. 17B. Each pair of long and short lines represents one cell or unit. The short line is always the — terminal of that cell, and the long line is +.

The A battery in the circuit of Fig. 17A has only one cell, so we show only one long and one short line for it. (Of course, in other tube circuits, the A, B and C batteries may have entirely different numbers of cells.) The C battery which we are using as an example has three cells, so we use three pairs of lines side by side to represent it in Fig. 17B. Notice how there is a long line at the + end of the C battery, and a short line at the -- end.

The B battery in this example actually has thirty cells. It isn't necessary to draw in thirty pairs of lines on our diagram, however. We just let the B battery symbol have a few more pairs of lines than the other batteries have, and specify the size of the B battery in some other way (such as by writing its voltage on the diagram.)

Even' battery symbols are unnecessary and often omitted. It is entirely sufficient to label the circuit terminals which should go to voltage sources, as shown in Fig. 17C. Furthermore, even the lettering A— and A+ on the filament leads is often omitted when a tube has both filament and cathode, for then the filament connections can be interchanged without affecting performance. Just remember that the notations A—, A+, B—, B+, C—, and C+ on radio circuit diagrams indicate connections which are to be made to A, B and C voltage sources.

To secure practice in making battery symbols, copy the diagram in Fig. 17B several times freehand. Don't worry about making the short lines thicker than the long lines, and don't worry



FIG. 18. This free-hand pencil sketch is typical of diagrams made by radio men.

if your symbols look uneven. The sample sketch shown in Fig. 18 is certainly no marvel of perfection, yet it would be perfectly clear and acceptable to any radio man.

How Tubes Amplify. Both the plate voltage and the grid voltage affect the amount of electron flow through a three-electrode vacuum tube. Since the grid is closer than the plate to the cathode, the grid has much more effect than the plate upon the electrons which are coming off from the cathode. This means that a small variation in the grid voltage will change the electron flow through the tube far more than will that same variation in the plate voltage.

For instance, a 1-volt change in the grid voltage might have ten times greater effect upon electron flow than a 1-volt change in plate voltage would have. In other words, the radio tube in this example makes a 1-volt radio signal in the grid circuit have the same effect on electron flow as a 10-volt signal would have in the plate circuit. We then say that the tube is able to amplify a signal ten times.

Suppose we connected a microphone to the grid of a tube along with the C battery, as shown in Fig. 19. The audio signal voltage of the microphone will now alternately increase and decrease the total voltage which is acting between the grid and cathode. As a result, the electron flow from cathode to plate (the plate current) will vary in exactly the same manner as the microphone output voltage. Now if we place a suitable radio part in the plate circuit, we will find that the audio signal voltage across this radio part is many times stronger than



FIG. 19. Elementary diagram showing how a three-electrode radio tube can amplify the weak audio signal voltage of a microphone. Actual microphone connections would be quite different.

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It is not ordinarily practical, however, to connect a microphone directly to a radio tube; the connection is usually made through a simple device known as a *transformer*. Since a transformer is another highly important radio device, it will be worth while to consider briefly here its general characteristics.

Transformers

What Is a Transformer? A transformer in its simplest form is nothing more than two coils of wire mounted close to each other, with no wire connections whatsoever between the two coils. The coil across which we apply our original signal voltage (the *input* tails of its construction appear in Fig. 20B.

If there is only insulating material like paper or fiber inside the coils (no iron), we have what is known as an air-core transformer or a radio frequency transformer.



FIG. 20A. Typical iron-core transformer used in radio circuits.

FIG. 20B. Iron-core transformer partly disassembled, to show how the coils fit over the center leg of the iron core. Note that the core is built up from many thin sheets of iron.

signal voltage) is called the primary coil or primary winding. The coil from which we take the new signal voltage (the output signal voltage) is called the secondary coil or secondary winding.

The material on which the coils of a transformer are wound is known as the *coil form*. When the coils are wound one over the other on a paper form inside of which is a *core* of thin iron sheets, we have an *iron-core transformer*. A typical transformer of this type is shown in Fig. 20A, and de**Iron-Core Transformer.** When a current is sent through the primary coil of an iron-core transformer, the iron will become magnetized. If the electric current is varying, as it is in the case of an audio signal, the number of magnetic lines of force flowing through the iron core will vary in an exactly corresponding manner.

Since the iron core passes through both the primary and secondary coils of the transformer, the varying magnetic lines of force are also passing through the secondary coil. These changing magnetic lines of force cause the electrons in the secondary coil to move just like the electrons in the primary coil are moving, and the result is that a signal voltage is *induced* (created) in the secondary coil.

Notice the use of magnetism; the magnetic lines of force flow through the iron core and make it possible to transfer an electrical signal from one coil to another in an iron-core transformer. This is such an important action to a radio man that it is worth repeating here in a form which will be easy for you to remember.

Whenever we change the number of magnetic lines of force which pass through a coil, a voltage is induced in the coil.

If the secondary winding has the same number of turns of wire as the primary, the transformer merely transfers the signal without changing its voltage. For this condition, we use the symbol shown in Fig. 21A, where the two windings are drawn to have the same number of loops. The letter P identifies the primary winding, while S identifies the secondary winding.

The straight lines between the coil symbols indicate that the transformer has an *iron* core. The freehand way of drawing the iron-core transformer symbol of Fig. 21A is given in Fig. 21B; practice drawing this several times right now, if you have a pencil handy.

If the secondary has more turns of wire than the primary, the voltage induced in the secondary will be higher than the voltage fed to the primary, and we say that the voltage has been stepped up by the transformer. The symbol in Fig 21C indicates this by having more loops or curls in the secondary. The microphone transformer in the control room of a radio station is one example of a step-up transformer. If the secondary has fewer turns than the primary, the secondary voltage will be less than the primary voltage, and we say that the transformer has stepped down the voltage. The symbol in Fig. 21D is used to show this condition. The output transformer in a radio receiver (the one usually mounted right on the loudspeaker) is an example of a step-down transformer.

When radio men draw transformer symbols, however, they sometimes do not bother to show which winding has



FIG. 21. Symbols for iron-core transformers.

more turns. In many cases, they're interested only in showing that there is a transformer in the circuit. As you acquire radio knowledge and experience, you will gradually become able to tell which winding should have more turns, regardless of how the symbol happens to be drawn.

Insist Upon Understanding

Now that you have completed your study of this lesson in the logical manner set forth in the Study Schedule on the inside front cover, won't you pause for a moment? Think over what you've learned about radio in just this one lesson.

Already you have a clear idea of what the main parts in a broadcasting system are. You really do know a great deal about electrons now. The word magnetism means a whole lot more to you today than it did last month. Did you ever expect that you would learn so much about tubes and transformers in the very first lesson?

Yes, already you are well started on a clear, interesting and satisfying Course in radio. You will agree that there has been nothing difficult or "over your head" in this first lesson.

The wonderful thing about the N.R.I. Course is that the remaining lessons will be just as interesting as the first, even though you keep learning more and more from each one. Your knowledge builds up by leaps and bounds once you master the fundamental laws and facts given in the early lessons.

A worth-while practical radio education is worth every bit of the time and effort you give to secure it. Every time you tackle a new idea and master it, you achieve still another advance in skill and experience.

In this Course, you will supply the ambition, but we're right behind you, ready to give you help whenever it is needed. You will do the thinking and studying, but we're eager to help you—we're willing to explain anything in a different way whenever you desire. You know when you understand an explanation and when you don't. Make sure always that you do understand.

Do your very best to master each idea yourself, but don't ever expect that you'll be able to understand everything without some help. Sooner or later, you will come to something which just doesn't seem to make sense to you. Whenever this happens, just remember that we are prepared and eager to give you personal instruction and special explanations, if you'll tell us exactly what your difficulties are. You have a Technical Consultation Service blank for this purpose. We'll send you another consultation blank with our reply. Don't be afraid to use the blank you have when necessary, for we don't want any missing stones in the foundation you're building for a successful radio career.

Looking Ahead

In this lesson, you visited the six main parts of a radio system, and secured a "bird's-eye view" of how radio programs and messages are broadcast and received. Before studying radio parts and circuits in earnest, it is natural for you to expect a more complete picture—a real insight into the radio work you hope to do. This desire will be satisfied in the very next lesson.

You will learn exactly how sound is picked up by the microphone in a broadcasting studio and converted into an electrical signal, then boosted in strength by amplifiers and given a ride through space on a radio wave. Following the radio wave again to a far-off receiver, you will study the fascinating radio tube circuits which convert the radio signals back into sound again. After this little journey, you will be able to recognize many radio parts and tell what their jobs are in a radio system.

Turn back to the inside front cover now, to see what comes next in your Study Schedule.

Lesson Questions

Be sure to number your Answer Sheet 1FR-3.

Place your Student Number on every Answer Sheet.

SPECIAL NOTICE: Mail us your answers for this lesson immediately after you finish them. This will make sure that the graded answers reach you while the subject matter is still fresh in your mind, and you will get the greatest possible benefit from our speedy personal grading service. Never hold up a set of lesson answers.

- 1. What is an audio signal?
- 2. Name, from memory, at least five of the six important sections in a complete broadcasting system.
- 3. Will an electron (a *negatively* charged particle) be attracted by another *negatively* charged object? (Remember that all charged particles and objects must obey the Law of Electric Charges, which tells what like and unlike charges will do.)
- 4. Which terminal of a battery has more electrons than normal?
- 5. Will the current in a complete circuit increase, decrease or remain the same when you increase the voltage of the battery?
- 6. Will the strength of an electromagnet be reduced when the current through the coil is reduced?
- 7. What is the name of the electrode which attracts the electrons emitted by tto cathode in a radio tube?
- 8. Name the three operating voltages required by a three-electrode radio tube.
- 9. What flows through the iron core of a transformer and makes possible the transfer of electrical signals from the primary coil to the secondary coil?
- 10. Draw the simple SYMBOLS (not picture sketches) for the following radio parts:
 - A-Symbol for a radio tube having a filament, cathode, grid and plate.
 - B-Symbol for a battery having about six cells.
 - C-Symbol for an *iron-core transformer* having about the same number of turns on both primary and secondary.

10-30-47

HOW TO DEVELOP CONFIDENCE

Self-confidence—an active faith in your own power to accomplish whatever you try to do—is a personal asset which can do big things for you.

One thing which builds self-confidence is a successful experience. Each lesson completed with a passing grade is a successful experience which will build up confidence in you.

Little successes are contagious. Once you get a taste of success, you'll find yourself doing something successful every day. And before you realize it, your little successes will have built up to that big success in radio you've been dreaming of. So—get the habit of success as fast as possible. Resolve to study every day, even if only for a few minutes.

Another confidence builder is a deep, firm faith in yourself—in your ability to get ahead. If you do believe in yourself and you are willing to back up this faith with good hard studying, you can safely leave the final result to itself. With complete confidence, you can look forward to an early success in radio.

Act as if you could not possibly fail, and you will succeed!

J. E. SMITH.



A 20



STUDY SCHEDULE NO. 2

By dividing your study into the steps given below, you can master this part of your N.R.I. Course quickly and thoroughly. Check off each step when you finish it.

🗆 1. l	Read:—How This Lesson Will Help You	Page 1
□ 2 . I	Microphone in Studio Read once at an ordinary rate, then re-read slowly so you clea what sound is, how it is produced, how it travels, what frequence and how a microphone changes sound into audio signals. Answer Lo 1, 2 and 3 on page 29 now.	Pages 2-8 rly understand ties it includes, esson Questions
3. 0	Control Room Read the first time at your usual reading speed, then re-read slov clear understanding of what an audio signal is and how the strey audio signal is boosted by the audio amplifier in the control of Lesson Question 4 now.	•.Pages 8-13 why to secure a ngth of a weak coom. Answer
□ 4. ′	Transmitter This is information which is included to complete your picture of Let your own interests determine how much time to spend on th thorough reading will ordinarily be sufficient. Answer Lesson Que	Pages 13-18 a radio system. his section; one stion 5 now.
□ 5 . [•]	Transmitting Antenna	Pages 18-20 u should, how- cesson Question
☐ 6 .	Receiving Antenna You can master this short section in a short time and get a satis standing of how the receiving antenna feeds signal currents do ceiver. Answer Lesson Question 7 now.	Page 20 sfactory under- own to the re-
□ 7 .	Radio Receiver. This is a highly important section both for servicemen and c men, so read it slowly at least three times. A clear general idea of tuned radio frequency receiver changes antenna signals into so you to master later lessons, because basic receiver circuits are to types of radio equipment. Answer Questions 8, 9 and 10 now.	.Pages 21-26 ommunications of how a simple ounds will help be found in all
₿ .	A Review. Read the review slowly several times to refresh your memory o portant points covered in this lesson. A single reading will be "Looking Ahead" section, which is a short discussion of the information you are going to get from the next lesson and how r already obtained.	.Pages 27-28 In the most im- enough for the practical radio much you have
9 .	Mail Your Answers for Lesson 2FR-5 to N. R. I. for Grading.	
□10 .	How To Use Radio Tools	ooklet No. 2 al, how-to-do-it andy for future lets is going to II want to read tudy Schedules ooklets. Read see Booklets are replacing them.
□11.	Start Studying the Next Lesson.	

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World Radio History

HOW RADIO PROGRAMS ARE SENT FROM THE STUDIO TO YOUR HOME

How This Lesson Will Help You

LEARNING radio is much like building a skyscraper. Once the steel framework of a skyscraper is in place, the various floors of a skyscraper

This lesson takes you through the same broadcasting system you studied in the first lesson. Now, however, you will go through more slowly, so



Courtesy RCA Mig. Co., Inc.

Courtesy General Electric Co.

A radio program starts at a microphone which is usually located in a broadcasting studio. On special occasions, however, this microphone may be set up at any point to which man can travel above, under or on the face of the earth. From this microphone, the program travels over a long and fascinating path to the loudspeakers of countless radio receivers. Somewhere along this path, there is a good-pay radio job awaiting completion of your training.

can be finished in any desired order. Likewise, once you become thoroughly familiar with the various radio parts and circuits which make up a complete modern broadcasting system, you will be in a position to study these radio parts and circuits faster.

Yes, our experience gained from many years of teaching radio has proved this to be true—that radio is easiest to master when you have a clear idea of how each study subject fits into the complete picture of radio. you have time to get acquainted with more of the *important* parts.

This detailed study of a radio broadcasting system will be your framework for the entire N.R.I. Course, regardless of what radio work you eventually choose. Even if you plan to follow radio receiver servicing or some other branch of radio dealing chiefly with receivers, a broad, general understanding of how radio programs are put on the air is essential to the completeness of your training.



1 MICROPHONE IN STUDIO. In radio, we start with sounds at the microphone, and must end up with reproductions of these same sounds at the loudspeaker. It is only logical, then, that you find out what sounds are like before you learn how microphones convert sounds into audio signals.

What Is Sound? When you strike a piano key, a small hammer swings against a wire inside the piano and makes that wire vibrate. If you touch the wire lightly, you will be able to feel these vibrations. Your ear can hear the vibrating wire even though many feet away, however, so these vibrations must travel through the air.

When you blow a whistle, the air inside and outside the whistle vibrates. Raindrops falling into a lake start many small vibrations both in the water and the surrounding air. The vocal cords in your throat vibrate when you speak or sing.

Since all these different vibrating materials produce sounds which can be heard, we can simply say that sound is a vibration of any material at a rate which can be heard by human ears. This vibrating material can be a solid object like wire, a liquid like water, or a gas like air.

How Sound Travels. Sounds produced by vibrating bodies must usually travel some distance before they reach human ears. Sound can travel through any gases, any liquids, and any solid objects which are elastic enough to vibrate. In other words, sound can travel through anything which possesses the ability to vibrate. Sound

cannot travel through a vacuum, because there is nothing in a vacuum which can vibrate. This and other interesting characteristics of sound are illustrated in Fig. 1 (on next page).

Most of the sounds which we hear travel through air, which is a mixture of invisible gases and dust particles. Of particular interest to us as radio men is the way in which a loudspeaker sends sound through the air.

How a Londspeaker Produces Sound. In an ordinary radio loudspeaker, we have a large paper cone which vibrates back and forth when the radio set is operating. This vibrating cone alternately pushes and pulls on nearby particles of air, thereby setting these air particles into vibration.

When the loudspeaker cone pushes air forward, it compresses the air particles, making them bump into other air particles which are farther away. These bump into still more particles, thereby giving a region of higher air pressure than normal. One of these higher-pressure regions begins traveling away from the loudspeaker each time the cone pushes forward.

When the cone moves backward, it leaves more room directly in front of itself for air particles, thus reducing the air pressure and creating a partial vacuum (a vacuum is a space having no air, and hence no pressure). Nearby air particles rush into this lower-pressure area, leaving a partial vacuum where they were. This lowerpressure region likewise travels outward in all directions from the cone of the loudspeaker.
Thus, a lower-pressure region begins traveling away from the loudspeaker each time the cone moves backward. This process repeats itself each time the cone moves forward and backward, which happens many times each second. The resulting regions of air pressure higher and lower than normal, traveling away from a sound source, are known as sound waves.

One thing we should definitely understand about sound is this: The vibrating particle of air which bumps into your eardrum and makes you hear the voice of a radio announcer did not travel all the way from the radio loudspeaker to you.

A sound wave made up of vibrating air particles can be compared to a wave moving across a lake on a windy day. This traveling wave is just a slow up-and-down vibration of water.

The water particles stay in practically the same locations all the time, and only the vibrating or up-and-down action of water travels across the lake with the wind. Likewise, air particles in a sound wave merely move back and forth without getting very far. Each particle transfers its backand-forth movement to the next particle, and these back-and-forth movements form traveling regions of higher and lower air pressure than normal.

Speed of Sound. Sound waves travel through air at a speed of approximately 1089 feet per second. This is slow in comparison to the 186,000-mile-per-second speed of radio waves, so you may be hearing the words of a radio announcer coming from your loudspeaker a split second before those sound waves reach persons at the rear in the studio audience.



FIG. 1A. The radio scientist says YES, because he knows shattered pieces of coconut will vibrate at a rate which can be heard by human ears. The medical scientist says NO, because he defines sound as a vibration acting on and audible to human ears, and there are none on this unin-habited South Sea isle

near the Equator.

FIG. 1B. No matter how violent an explosion of the moon might be, neither you nor any other person on this earth could hear the sound of the explosion. The reason is simply that sound travels only through materials which can vibrate, and there is nothing in the ethereal space between us and the moon-just VACUMR.

FIG. 1C. When a large drum is struck vigorously, the stretched diaphragm of the drum vibrates in and out. This action alternately increases and decreases the air pressure in the vicinity of the drum, creating a sound wave which travels away from the drum. Your hand definitely feels the varying air pressure of this sound wave.

FIG. 1D. If some husky youngster should blow a large whistle or horn close to your ear, you would definitely agree that sound can be painful to the eardrum if sufficiently This is why loud. artillery men must stuff cotton into their ears or use special ear plugs when firing big guns during land warfare or naval maneuvers.

How We Hear Sound. The outer passage of the human ear is closed by a thin membrane called the *eardrum*. When the eardrum is set into vibration by sound waves which travel through air and enter the ear, the vibration is transferred through a series of small bones to a fluid in the inner ear, and through this fluid to the sensitive fibers of the hearing nerves. These nerves convey to the brain the sensation of sound.

Wavelengths and Cycles. If your sense of touch were sufficiently delicate, you could stand in the path of a sound wave and feel every variation above or below normal air pressure, just as is illustrated in Fig. 1C. You would feel an *increase* in the pressure of the air, followed by a return to normal. a decrease from normal air pressure, and a return to normal again. This would repeat itself as long as the sound was being produced. Each complete increase and decrease in air pressure, corresponding to one complete vibration back and forth of the sound source, is known as a cycle of the sound wave (pronounced SIGH-kull).

If you could in some way stop a sound wave and make its variations in air pressure visible, you could take a yardstick and measure the distance between two adjoining higher-pressure regions of the sound wave. Your result would be the length of the wave, or the *wavelength* of that particular sound. Wavelengths of common sounds range from about one inch for the chirp of a cricket to about fifty feet for the booming sound of a bass drum.

Frequency of Sound. The number of complete back-and-forth vibrations which a sound source makes in one second of time is called the *frequency* of a sound. If you could count the number of vibrations the cone of a loudspeaker makes in one second, your

result would be the *frequency* of the sound in cycles per second.

The lowest sound frequency which a person can hear is about 20 cycles per second; the highest is about 20,000 cycles per second. The higher the frequency, the more high-pitched or shrill the sound is to our ears.

The approximate range of frequencies which we can hear is thus between 20 cycles and 20,000 cycles; this range is called the audible range. Any frequency in this range is an audio frequency, abbreviated a.f. Of course, these figures represent the extreme limits of hearing, and the average person will have a somewhat narrower hearing range. Older persons, especially, have difficulty in hearing the higher audio frequencies.

Notice that we omitted the phrase "per second" after the frequency values in the previous paragraph. Radio men often do this, and no one is confused, simply because the only interval of time ever used with frequency in radio work is one second. Just remember that whenever you read "20 cycles" or any other value in cycles, the meaning is "cycles per second," unless some other meaning is definitely indicated.

Program Radio Frequencies. Most of the radio broadcasting systems in this country are designed to handle audio frequencies from about 80 cycles to about 5000 cycles. A few high-fidelity broadcasting stations handle frequencies up to about 10,000 cycles, while the new frequency modulation broadcasting systems (developed by Major E. H. Armstrong, and described later in the N.R.I. Course) handle audio frequencies from about 30 cycles to about 15,000 cycles. Increased frequency range makes programs seem more natural when heard with frequency modulation receivers.

Audio Reminders. You will encounter the terms audio and a.f. hundreds of times throughout this course. Just remember that they both mean the same in common usage, and both refer to frequencies which can be heard.

For example, take the audio signal in a receiver. It is strictly an electrical signal, and cannot be heard. Because it is an audio signal, however, we know it has the same frequency (the same number of cycles per second) as the original sound. By feeding this audio signal into a loudspeaker, we can make the nearby air vibrate at exactly this same frequency, and we can hear the resulting sound waves.

The Microphone

As you have already learned, the purpose of a microphone in a radio system is to change sound waves into audio signals. The microphone does this job in two steps: 1. It changes the sound waves into corresponding backand-forth movements or vibrations of a flexible part; 2. It changes these vibrations into audio signals.

How Sound Waves Move a Diaphragm. Every microphone has one flexible part which vibrates whenever sound waves hit it. This part is called the *diaphragm*, and is usually a thin, light-weight metal disc which is loosely mounted so it can be pushed back and forth by sound waves.

With the aid of a pencil, a length of paper tape, a telephone mouthpiece and a few small parts arranged as shown in Fig. 2A, we can demonstrate how the diaphragm of a microphone acts when a sound wave comes along. (In this diagram, half of the mouthpiece is cut away to show the diaphragm.) The diaphragm and pencil are joined by a link and lever in such a way that the pencil moves to the left when the diaphragm is *pushed* by the increased-pressure part of a sound wave, and the pencil moves to the *right* when the diaphragm is *pulled* forward by the reduced-pressure part of a sound wave.

If we let the point of the pencil rest on a length of paper tape, and pull this tape upward at a uniform speed while the sound wave of a pure whistled note is making the diaphragm vibrate, the pencil will trace





on the paper a smooth wavy line like that appearing on the tape in Fig. 2A.

If the sound wave is produced by a man saying "ah-h-h-h," however, the pencil will trace on the paper a jagged pattern like that shown in Fig. 2B. Each other type of sound will give a different pattern on the paper tape, and each pattern will represent the way in which the diaphragm and the source of sound are vibrating.

The one important thing for you to remember is that whenever sound waves hit the diaphragm of a microphone, they make this diaphragm vibrate in the same way that the original source of sound is vibrating.

How a Diaphragm Produces Audio Signals. Although all microphones start out by converting sound waves into vibrations of a diaphragm, many different types of microphones are used for changing diaphragm movements into audio signals. The important types in use today are the dynamic microphone, crystal microphone and velocity microphone. Condenser microphones and carbon microphones were once popular, but are now gradually fading out of the radio picture.

Let us assume that the microphone being used in the studio we are inspecting is a *dynamic microphone*. This is today one of the most widely used of all microphones, so an understanding of how it works will be sufficient until you come to the N.R.I. lesson which covers the other types of microphones.

The important features of a dynamic microphone are shown in a simplified manner in Fig. 3. Sound waves enter the microphone through the holes in front, and make the thin circular metal diaphragm vibrate in the same way as the original sound source is vibrating.

Attached to the back of the diaphragm is a paper-thin bakelite coil form about half an inch in diameter. On this cylindrical form is a small coil of wire called the voice coil, made from insulated copper wire almost as fine as human hair. The voice coil has two flexible wire leads (pronounced leeds) which connect to the two wires of the microphone cable. The moving system of this microphone (the diaphragm and voice coil) is so light in weight that it vibrates readily even for faint sounds.

Surrounding the voice coil in Fig. 3 are the poles of a powerful permanent magnet, arranged in such a way that magnetic lines of force pass through this tiny coil at all times. Yes, that's all there is to a dynamic microphone —just a coil of wire mounted on a diaphragm, surrounded by a magnet, and connected to the two wires of the microphone cable.

When the diaphragm of a dynamic microphone is set into vibration by a sound wave, the voice coil moves in



FIG. 3. Cross-section diagram of a typical dynamic microphone, showing the essential features which would be visible if we sliced vertically through the center of the microphone. When the voice coil of a dynamic microphone is moved in and out by sound waves, the number of magnetic lines of force which pass through the coil is changed, and therefore an audio voltage is induced in the coil. (The actual microphone appears in Fig. 6.)

and out through the magnetic lines of force which exist between the poles of the permanent magnet. This movement of the voice coil changes the number of magnetic lines of force which pass through the coil, thereby inducing (producing) an audio voltage in the coil.

When the voice coil moves into the magnet, the voltage acts in one direction; when the coil moves out of the magnet, the voltage acts in the opposite direction. The voltage which the microphone puts out (called the *microphone output voltage*) thus reverses its polarity (direction) from instant to instant. A voltage which regularly reverses or alternates its polarity in



FIG. 4. Patterns like these, traced on the screen of a cathode ray oscilloscope by a flying beam of electrons, show exactly how an audio voltage is varying. The weird, bright green fluorescent glow of the lines in these patterns is truly fascinating. (A fluorescent screen is one which glows when hit by electrons.) By using an oscilloscope to show how invisible voltages and currents are acting, radio engineers can pry into the secrets of radio circuits and actually see how adjustments or new parts affect performance.

this manner is known as an alternating current voltage (abbreviated a.c. voltage).

The a.c. voltage produced by a microphone has exactly the same *frequency* as the original sound, because the voltage reverses in polarity each time the original sound source reverses its direction of motion. For this reason, the a.c. voltage which is generated in the voice coil of a dynamic microphone is generally called the audio voltage or *a.f. voltage*. Sometimes this audio voltage is simply called the *audio signal* or *a.f. signal*.

The louder the sound, the greater is the movement of the voice coil and the stronger is the audio voltage produced by the microphone.

Audio Voltage Patterns. When radio men want to know how a particular audio signal voltage is varying, they sometimes use an electrical instrument called the *cathode ray oscilloscope* (pronounced *ah-SILL-oh-SKOPE*). In this instrument, the varying voltage is made to move a beam of electrons up and down across the white screen at the large end of a cathode ray tube. A bright greenishwhite glow appears wherever the electron beam hits the chemical coating which forms this *fluorescent* screen (pronounced *FLEW-oh-RESS-sent*). A fluorescent screen glows *while* being hit by electrons.

When the audio signal voltage applied to an oscilloscope is zero (or when there is no sound), the electron sweeps horizontally from side to side continually, and "paints" on the screen the horizontal line shown in Fig. 4A.

When we use an oscilloscope to look at the audio voltage produced by a pure whistled note, the audio voltage pattern shown in Fig. 4B appears on the screen. Compare this with the pattern produced by the moving pencil in Fig. 2A, and you will see that this audio voltage varies exactly in step with the movements of the microphone diaphragm. The horizontal line still shows faintly here, for the electron beam returns along this line while tracing the pattern over and over again. Each time the pattern crosses the horizontal line, the voltage drops to zero and reverses its polarity.

When we apply to an oscilloscope the audio voltage produced by a microphone while a man is saying "ah-h-h," the audio voltage pattern shown in Fig. 4C appears on the screen. Comparing it with Fig. 2B, we find again that the a.f. voltage of a microphone varies in step with the diaphragm movements.

The audio voltage produced by a microphone makes electrons move back and forth through the microphone wires when a complete circuit is provided for the microphone by apparatus in the control room.



2 CONTROL ROOM. The audio signal voltage delivered by a microphone is so weak that it must be boosted a great deal before it can be sent over telephone lines to the transmitter. This building up of the signal voltage is done in the control room, under the watchful eye of the monitor operator. The control room of a typical radio station is pictured in Fig. 5. Let's imagine that we have just stepped into this room, and have asked the monitor operator to tell us about his work. His answer might run something like this:

"The audio signal travels through the microphone cable to this soundproofed wall in front of me. The sig-



Courtesy Western Electric Co.

FIG. 5. The monitor man in the control room watches the performers in the studio through a large glass window as he "rides gain." The small microphone at the left of the control unit is used principally for intercommunication between the monitor room and studio prior to a broadcast or during rehearsals, for the wall between the two rooms is sound-proof.

8

nal is brought through the wall by, program is getting too loud, I immemeans of a plug and outlet device on each side, after which it travels through another cable to this audio amplifier unit on my desk. program is getting too loud, I immediately cut down the volume, to prevent overloading of the transmitter circuits. Overloading destroys the naturalness of the sound signal, and



FIG. 6. The first circuit in our studio-to-home radio system extends from the microphone in the studio to a microphone transformer in the audio amplifier which is located on the control desk in the control room.

"Of course, there are other microphones in the studio, each feeding into my amplifier over a similar cable, plug and outlet path. Switches allow me to change instantly from one microphone to another during a broadcast, or use any combination of microphones when the program includes a dramatic production or music by a large orchestra.

"This heavy dual plate-glass window, built into the wall between the control room and the studio, blocks all sounds but allows me to watch the artists and the announcer while I 'ride gain' on the program. This is my most important duty; to 'ride gain' means to adjust the volume control here on the panel continually to compensate for excessive changes in the loudness of the program.

"For example, if I notice that the announcer is gradually moving away from the microphone, I gradually advance this volume control so the listeners cannot notice any change in the loudness of their program. If this meter in front of me indicates that the program is getting too loud, I immediately cut down the volume, to prevent overloading of the transmitter circuits. Overloading destroys the naturalness of the sound signal, and can even put the transmitter 'off the air' for a few moments because of the operation of automatic circuit-breaking switches which protect tubes and parts from burning out during overloads.

"But let's follow the audio signal as it enters this amplifier. The first thing it encounters is an iron-core transformer known as the *microphone transformer*. Notice that the two microphone wires go to the two terminals on the primary winding on this transformer (see Fig. 6); this gives us a complete closed circuit over which electrons can travel.

"Here's the way I like to imagine that electrons are moving through the microphone circuit. I think of a neck-



Using a cathode ray oscilloscope to see how the a.f. signal voltage in a radio set is varying.

lace of balls strung end to end around the entire circuit, as on this diagram (see Fig. 7). The lever of our microphone diaphragm mechanism (the same mechanism as in Fig. 2) is attached to one point on the necklace. "Now, when a sound wave pushes down on the diaphragm, it makes the necklace end of this lever move upward, and the whole string of balls has to travel in the direction indicated by the arrows in Fig. 7.

"When the sound wave has pushed the diaphragm as far down as it can, both the diaphragm, the lever, and the necklace stop moving. Then, as the diaphragm moves backward due to its own springiness and to the pulling effect of the following lower-pressure start up the car gradually as you leave one point, you travel at maximum speed when half-way between the two points, and you start slowing down before you arrive at the other point so as not to go past it.

"The actual electrons which make up an alternating current or signal current in a radio circuit move back and forth just like this imaginary necklace of balls. Of course, a piece of wire has millions of times as many electrons as there are balls in the



FIG. 7. To picture the way in which electrons move back and forth in a wire to make up an audio signal, think of a necklace of balls which is being moved back and forth through a circuit.

region in the sound wave, the lever starts moving downward, and the entire necklace moves in the opposite direction to that shown by the arrows in Fig. 7.

"We thus have our necklace starting up gradually in one direction, increasing rapidly in speed, slowing down again as the diaphragm reaches the limit of its travel, stopping as the diaphragm stops, then reversing in direction and again repeating this increase and decrease in speed.

"You can compare the speed of this necklace to that of an automobile which you are driving back and forth continually between two points. You necklace shown in Fig. 7, and there are no strings connecting the electrons together. But—whenever the electrons in a wire are acted on by an a.c. voltage, the electrons just move back and forth without getting anywhere, like the necklace of balls. That's all anv audio signal *current* is—just a back-and-forth movement of electrons through the wires and parts of a cir-The lower the frequency, the cuit. back-and-forth movements fewer there are per second.

"Here is a simplified circuit diagram which represents this first stage or section of our radio system (see Fig. δ now). The audio signal current which flows through primary winding P of the microphone transformer causes an audio signal voltage to be induced in secondary winding S of this transformer. This audio voltage is several times stronger than the microphone voltage, because the microphone transformer secondary winding has more turns of wire than the primary.

"Notice that the stronger audio signal voltage which exists in secondary S acts upon the grid and cathode of the first a.f. amplifier tube, along with the d.c. voltage of the C battery which the time. Each circuit is so planned, however, that the motion of the electrons in one circuit controls the motion of the electrons in the next circuit in the desired manner.

"To make this perfectly clear, let me redraw the circuit of Fig. ϑ in such a way that each complete path for electrons is separate from the others (see Fig. ϑ).

"The electrons which travel back and forth in the microphone circuit cannot get into the grid circuit . . . cannot get into any other tube circuit



FIG. 8. Simplified circuit diagram showing the first part of the path taken by signals in traveling from the studio to your home. Only the most important parts of the circuit are shown here; the complete circuit will be taken up in later lessons.

is in this same circuit. This first tube and its B battery together have amplifying ability, so a still stronger a.f. voltage is delivered to the primary of the second transformer in our circuit.

"A microphone transformer and an amplifier stage acting together may amplify an a.f. signal voltage several hundred times, but even this is not enough. We must have many stages in this amplifier, with many tubes and transformers. Each stage steps up the signal voltage some more. Finally, we secure a strong enough voltage to send our signal over the telephone lines to the transmitter.

"There is just one more thing I must make clear before you leave the control room. Electrons do *not* move from one circuit to another in a radio system. The electrons in motion in a circuit remain in that circuit all ... and obviously cannot ever get into the transmitting antenna. Only magnetism, acting in the microphone transformer, links the microphone circuit with the grid circuit. It is magnetism which induces in our grid circuit the audio signal voltage which acts on the grid and cathode of the tube.

"You might at first think that the C battery would produce a continuous movement of electrons in one direction through the first grid circuit in Fig. 9, but a little study will show why this does not happen.

"First of all, we learned in the previous lesson that the *negative* terminal of a battery tries to send electrons through the circuit to the *positive* battery terminal. With the C battery connected as shown, its negative terminal will try to make electrons flow from the grid to the cathode through the tube, so they will get to the + terminal of the C battery. No electrons can flow this way, however, because the grid does not emit any electrons. Electrons emitted from a heated cathode are the only ones which can bridge across the gaps between the electrodes of ordinary radio tubes.

"This is an idea well worth remembering, so let's repeat it in simpler form: When the grid of a vacuum tube is *negative*, there can be *no continuous* flow of electrons in the grid circuit.

"The electrons in the plate circuit are influenced by the grid circuit electrons, but there is no movement of electrons between the two circuits through the tube. We might think of these grid circuit electrons as traffic officers stationed on the grid wires of the tube, regulating the electron traffic which passes from cathode to plate in the plate circuit.

"Thus, you can see how a few electrons at the grid of a tube can control a much greater number of electrons in the plate circuit. In other words, a weak audio signal at the grid gives us a much stronger audio signal in the plate circuit. The stronger signal will behave exactly like the weak signal that is, will have the same frequency.

"The B battery in the plate circuit

is connected to make the plate positive with respect to the electron-emitting cathode, so the plate attracts electrons and we have a continuous electron flow from the cathode to the plate and around the complete plate circuit—a d.c. plate current.

"We thus have both a d.c. plate current and an audio signal flowing through the plate circuit at the same time. There is nothing unusual about this; the audio signal voltage in the grid circuit simply makes the d.c. plate current fluctuate above and below its steady value, and these momentary changes in the amount of electron flow (alternate increases and decreases) form the audio signal in the plate circuit.

"The B battery provides the additional power needed to make the audio signal in the plate circuit stronger than the signal in the grid circuit. In fact, no tube can generate electrical power; it merely utilizes the power from a voltage source in such a way as to amplify or strengthen a signal.

"It is this audio signal flowing through the primary of the second transformer (Fig. 9) which induces an audio signal voltage in the secondary of the transformer (in the second grid circuit).





"Each other circuit in this audio amplifier likewise has its own set of electrons and its own audio signal. Tubes and transformers make elec-

trons in one circuit act upon electrons in the next circuit, and thus we secure the desired increasingly stronger audio signal in each succeeding circuit."



3 TRANSMITTER. We leave the monitor operator in his control room, and follow the audio signal through the telephone lines which run to the transmitter.

By the time the audio signal has arrived at the transmitter, its voltage (strength) has dropped considerably due to the long trip through the telephone line. This is why the audio signal is sent through several more voltage-boosting a.f. amplifier stages after it reaches the transmitter.

R.F. Carrier Signal. No matter how strong we make the audio signal voltage, however, it will not produce radio waves when fed into the transmitting antenna. The only thing that will produce radio waves which will "go places" is a high-frequency signal. This required higher frequency is known as a radio frequency, and is abbreviated r.f. A radio frequency signal is usually higher than 100,000 cycles per second, whereas, you will remember, an audio frequency signal is rarely higher than 15,000 cycles.

A radio frequency signal can be used to carry an audio signal through space. When used in this way, the r.f. signal is called the r.f. carrier signal. Each radio station in the United States is assigned a definite carrier frequency value by the Federal Communications Commission.

Crystal Oscillator. The r.f. carrier signal for a modern broadcast transmitter is produced by a special radio vacuum tube circuit known as a crystal oscillator. This circuit contains three important radio parts which we have not yet taken up a crystal, a variable tuning condenser and an r.f. transformer. It is the crystal which produces the r.f. carrier signal, so we will first see what the crystal is like.

The Crystal. A modern radio transmitter is kept on its assigned carrier frequency by a carefully ground piece of natural quartz crystal about one inch square, which is known to radio men as a *crystal*. One of these finished crystals is shown in Fig. 10, alongside a block of natural quartz from which radio crystals of



Courtesy General Electric Co.

FIG. 10. A quartz crystal slab which is ready for use in a broadcast transmitter is here held by tweezers alongside a piece of quartz as found in its natural state. this type are cut. Natural quartz is mined in South America, in Brazil.

Frequency of a Crystal. When a piece of quartz crystal is jarred or tapped, it will vibrate like a strip of stiff spring steel. Each crystal has its own natural frequency of vibration, which depends upon the thickness of the crystal. For example, a crystal one-fifth of an inch thick has a natural vibration frequency of about 600,-000 cycles.

A vibrating crystal has another peculiar characteristic—it produces an a.c. voltage between its faces whenever it vibrates. This a.c. voltage has the same frequency as that at which the crystal is vibrating. Still more interesting is the fact that we can reverse this characteristic—we can make the crystal vibrate by connecting it to an a.c. voltage source.

Crystal Oscillator Circuit. A simplified version of a crystal oscillator circuit is shown in Fig. 11. The crystal is connected between the grid and cathode of the tube, while the B battery and a *tuning circuit* are connected between the plate and cathode.

The coil in the tuning circuit is the primary winding of an r.f. transformer, which consists simply of two coils of wire wound side by side or one over the other on a length of paper or fiber tubing called a *coil form*. There is no iron core inside the coil form.

The variable condenser consists of two sets of metal plates, insulated from each other. One set of plates is fixed. The other set is mounted on a shaft which turns or rotates in such a way that its plates can be meshed any desired amount with the fixed plates. (Two sets of plates are meshed when they fit between each other without touching.)

When the crystal oscillator circuit of Fig. 11 is operating properly, the crystal is vibrating at its natural frequency. Since this is generally higher than 100,000 cycles, it is a radio frequency. The resulting weak r.f. voltage produced by the vibrating crystal acts between the grid and cathode of the tube. The tube and its B battery boost the strength of this r.f. signal, just as in an audio amplifier stage, so we have a stronger r.f. signal voltage in the plate circuit.

Some of the r.f. signal voltage in the



FIG. 11. Simplified diagram of a crystal oscillator circuit, with illustrations of the three parts you have not yet studied. Only the more important parts are shown, because the other parts will be taken up in later lessons. Thus, by-pass condensers are omitted, and no C supply is shown. The condenser is a *tuning* condenser.

plate circuit feeds back through the tube and gives the crystal a little "push" during each cycle of vibration. A small part of the power obtained from the B battery thus serves to keep the crystal vibrating. The explanation of how the signal feeds back comes later in your Course.

By rotating the shaft of the variable condenser, the tuning circuit can be adjusted for best possible operation of the crystal oscillator. Maximum r.f.



FIG. 12. All the main sections of a broadcast transmitter are shown here, along with the basic connections to the transmitting antenna.

current then flows through the coil of the tuning circuit, and this current induces a maximum r.f. voltage in the secondary of the r.f. transformer, for transfer to the next section of the transmitter. This r.f. voltage is still much too weak for broadcast purposes, however, so this r.f. carrier signal must be sent through a number of r.f. amplifier stages to build up its strength.

R. F. Amplifier. At this point, a "picture" of the main sections of a complete transmitter will help you to understand how the audio signal and the r.f. carrier get together. This picture is given in the block diagram in Fig. 12.

You will recognize the crystal oscillator and the a.f. amplifier immediately, for you have already studied these sections. The modulated stage is the highly important one which combines the audio signal and the r.f. carrier signal to produce the modulated r.f. carrier signal. The r.f. amplifter, located between the crystal oscillator and the modulated stage. boosts the r.f. output voltage of the crystal oscillator. The modulated r.f. amplifier is the most interesting of all sections, for it contains huge air or water-cooled output tubes which strengthen the modulated r.f. carrier signal sufficiently so it can be fed over the transmission line and through the dog house to the huge steel transmitting antenna tower.

Even in a moderate-size broadcasting station, one or more vacuum tube stages are required in the r.f. amplifier. These are all essentially alike, so only the first r.f. amplifier stage is shown in Fig. 13. Its operation is the same

CRYSTAL OSCILLATOR

R.F. AMPLIFIER STAGE





as that of an a.f. amplifier stage, except that an r.f. voltage instead of an a.f. voltage is applied between the grid and cathode of the tube and causes the plate current to vary at an r.f. rate. Remember that it is the B voltage source which furnishes to each stage the electrical power required to boost the signal strength.

The r.f. plate current flowing through primary P of the r.f. transformer in Fig. 13 induces a strong r.f. voltage in secondary S, which is in the grid circuit of the next r.f. amplifier r.f. carrier signal and transfers it also to the grid circuit of the modulated stage.

We thus have an r.f. carrier voltage and an audio voltage acting on the grid of the tube in the modulated stage, in addition to the usual C battery voltage. As a result, we have in the plate circuit a modulated r.f. plate current an r.f. plate current which increases and decreases in strength in step with the audio signal.

This signal-combining action is so important in a broadcast transmitter



FIG. 14. Simplified circuit diagram of the modulated stage in a radio transmitter. This stage places the audio signal on the r.f. carrier, so as to give a modulated r.f. carrier signal.

stage. The final r.f. amplifier feeds its r.f. voltage into the grid circuit of the modulated stage.

Modulated Stage. The next stage we are to investigate is known as the modulated stage, because in this stage the r.f. carrier is modulated (made to vary) in the same way that the audio signal is varying. As you can see from the diagram in Fig. 14, the modulated stage contains only one more part than is in an r.f. amplifier stage. This part is an iron-core audio transformer like those you have already studied. The audio transformer transfers the audio signal from the audio amplifier to the grid circuit of the modulated stage, while the first r.f. transformer in Fig. 14 takes the

that it is well worth repeating as a single easy-to-remember sentence:

The modulated stage of a transmitter combines the audio signal with the r.f. carrier signal to give the modulated r.f. carrier signal.

With a cathode ray oscilloscope, the laboratory man can show exactly how each of the signal voltages in the modulated stage is varying. This gives a clearer idea of how the audio signal "rides" on the r.f. carrier, for we see an action which otherwise would have to be imagined because it is invisible.

In Fig. 15A we see the pattern for the signal voltage which is fed into the modulated stage. It is much like that shown in Fig. 2B for the "ah-h-h-h" sound.



FIG. 15. These three patterns give the complete story of the modulated stage in a transmitter, as it would appear on the screen of a cathode ray oscilloscope.

In Fig. 15B is the pattern for the r.f. carrier voltage which is also fed into the modulated stage. Its loops are close together because, as you will remember, this r.f. voltage reverses its polarity many more times per second than does an audio voltage.

The most interesting pattern of all is Fig. 15C, which shows how the modulated r.f. carrier voltage, which we obtain from the modulated stage. is varying in strength from instant to The farther the pattern instant. swings above and below the horizontal line at a particular instant, the greater is the strength of the signal at that instant. The pattern of our audio voltage is clearly recognizable both above and below the horizontal zerovoltage line, proving definitely that our radio program is still present at the output of the modulated stage. Thus do electrons in a cathode ray oscilloscope tube make it possible for us to find out what electrons in other circuits are doing.

The modulated r.f. plate current flows through primary P of the second r.f. transformer in Fig. 14, and induces in secondary S the desired modulated r.f. voltage. At last we have an audio signal "riding" on an r.f. carrier—the required combination for producing radio waves which will carry radio programs for long distances through space.

In low and medium-power transmitters, the modulated r.f. carrier signal is often fed directly to the transmitting antenna without additional amplification.

In large transmitters, however, the modulated signal is usually fed into a modulated r.f. amplifier. This contains one or more amplifier stages much like the r.f. amplifier stage you have already studied, except that now we are using much larger tubes.

Our signal acquires more power (more strength) each time it passes through an amplifier. In a large broadcast transmitter, the huge vacuum tubes in the modulated r.f. amplifier stage handle so much signal power and take so much power from the B voltage source that the tubes must be cooled either by blower fans or by a circulating water system. This is particularly true of the final tubes in the transmitter; these must handle the government-assigned output power of the transmitter, which for most U.S. broadcast stations is some power value between 250 watts and 50,000 watts. (A watt is a unit used for measuring quantities of electric power. The antenna of a 100-watt station gets the same quantity of power as does a 100-watt lamp bulb.)



TRANSMITTING ANTENNA. The modulated r.f. carrier signal at the output of a radio transmitter is fed to the transmitting antenna tower over a special two-wire transmission line which runs through the tuning house or "doghouse," essentially as was illustrated in Fig. 12. One of the transmission line wires goes to the base of the huge steel antenna tower. This tower is insulated (electrically separated) from the earth by means of insulators. The other transmission line wire goes to a network of buried copper wires which radiate outward for hundreds of feet from the tower like the spokes of a wagon wheel. These wires serve as the ground connection.

The transmitting antenna tower and the nearby ground area together act like a tuning circuit containing a coil and condenser. This means that maximum signal current will flow through the antenna tower when the tower is tuned exactly to the r.f. carrier frequency of our transmitter.

An antenna can be tuned to a definite frequency simply by adjusting the height of the antenna tower. Engineers figure out the height which will approximately give the desired tuning, then build the tower and provide at its top a telescoping metal pipe or mast which can be adjusted to increase or decrease the height of the antenna.

The tuning house or doghouse may contain an r.f. transformer with a condenser placed across each of its coils so as to provide two additional tuning circuits. These insure that the antenna will receive the maximum possible amount of signal power from the transmission line.

Remember, our modulated r.f. carrier signal in the transmitting antenna is simply a combination of our desired audio signal and its r.f. carrier.

Producing Radio Waves. In the previous lesson, you learned that whenever electrons move through a wire, they produce a magnetic field around the wire.

In addition to this, electrons have their own electric fields, consisting of electric lines of force which represent the directions in which electrons will repel or attract other charged objects.

When the electron flow through a transmitting antenna changes in amount or direction, both the electric and magnetic fields of the electrons will travel out in all directions from the tower at the speed of light. These moving electric and magnetic fields make up the electromagnetic waves which are known as radio waves.

During intervals of silence when no audio signal is produced by the microphone in the studio, the only signal which reaches the transmitting antenna is the r.f. carrier alone. We call it the *unmodulated* r.f. carrier signal, because it has no modulation (no audio signal riding on the r.f. carrier). When an unmodulated r.f. carrier signal is being supplied to a transmitting antenna by a broadcast transmitter, the electron flow in the antenna tower is changing in direction and amount many hundred thousand times per second, essentially as indicated by the r.f. carrier voltage pattern in Fig. 15B. As a result, electric and magnetic fields which also vary in much this same way are being produced continuously by the transmitting antenna. These electric and magnetic fields travel off into space as radio waves.

Here is the important fact to realize —only the magnetic and electric fields leave the transmitting antenna. The electrons stay right in the antenna tower at all times.

When the microphone is picking up sound waves and producing audio signals, the r.f. carrier signal is modulated with an audio signal. The maximum strength of the electric and magnetic fields then varies from instant to instant much as in Fig. 15C, because the electron flow in the antenna tower is now varying in strength in this manner. A radio wave therefore varies in strength from instant to instant in such a way that it still carries the same radio program we traced through the transmitter stages.

When radio waves leave a transmitting antenna tower, they travel up into the sky as sky waves, and along the ground as ground waves. Those waves which go up into the sky are not lost, however; about 50 miles above the earth there are layers of electrons and gas particles which can bend or reflect radio waves back to the earth.

All long-distance radio reception, in short-wave bands as well as the broadcast band, is by means of sky waves. Ground waves give the best reception, but have a range of only about 100



Courtesy Chicago Tribune & WGN.

Courtesy Station WFOY.

Left: Famous 750-foot high steel transmitting antenna of radio station WGN in Chicago. A single porcelain insulator supports the 65-ton weight of the tower.

Right: Self-supporting 200-foot high steel tower of station WFOY, located in a salt-water marsh near St. Augustine, Florida. The dog house (tuning house) is directly under the tower, with a catwalk running from it to the transmitter. miles at the most. Transmitting antennas for broadcast and police radio stations are designed to send out a high proportion of ground waves, while antennas for short-wave stations are designed to send out chiefly sky waves. This completes the first chapter of our story of radio; in the next chapter, we discover how the receiving antenna and receiver reverse the entire transmitting process and give us sound waves again.



5 RECEIVING ANTENNA. When a radio wave arrives at a receiving antenna, the electric and magnetic fields of this radio wave make the free electrons in the antenna wire move back and forth *along the wire* at exactly the same frequency as that of the r.f. carrier in the transmitter. One electron transfers its motion to the



FIG. 16. Simple receiving antenna system, showing path from antenna to ground for modulated r.f. currents set up in the antenna by radio waves. Symbols used by radio men for the antenna and ground are shown at the right. next through collision, so the electrons in the receiving antenna move back and forth in the same way as the electrons in the transmitting antenna. The result is a modulated r.f. current in the receiving antenna, like the current in the transmitting antenna but much weaker. The farther the radio waves have to travel, the weaker is the receiving antenna current.

The circuit for this modulated r.f. current is from the antenna down to the receiver, through the first r.f. coil, and then to the ground, as shown in Fig. 16.

The farther the receiving antenna is from the transmitting antenna, the weaker will be the radio wave by the time it reaches the receiving antenna, and the weaker will be the resulting modulated r.f. current in the antenna.

Each receiving antenna has a natural frequency of its own, at which the antenna current is the strongest for a given radio wave. Changing the length of the antenna changes its "best-results" frequency.

For ordinary broadcast band reception of entertainment programs, we pay little or no attention to the length of a receiving antenna. It is only when best possible reception of distant short-wave stations is desired that we begin measuring antenna lengths.



RECEIVER. Two types of re-6 ceivers are in general use today. The simpler of these is the *tuned radio* frequency receiver, commonly referred to as a t.r.f. receiver. The other type is the superheterodyne receiver, which is now the leader because its special circuits give superior results. Naturally you will prefer to take up first the simpler of these receivers, so we will consider only t.r.f. receivers now.

As the diagram in Fig. 17 indicates, a t.r.f. receiver has only four sections. having the following purposes:

1. The r.f. amplifier, which builds up the strength of the signal from a desired station, and also keeps out undesired signals.

2. The detector, which separates the audio signal from the r.f. carrier.

3. The audio amplifier, which builds up the strength of the audio signal.

4. The loudspeaker, which converts

Fig. 17 he says immediately that our t.r.f. receiver has two r.f. amplifier stages, with three tuning circuits. The two r.f. amplifier stages boost the strength of the desired modulated r.f. carrier signal a tremendous amount. The three tuning circuits keep back all other modulated r.f. carrier signals which may be present in the receiving antenna, and also do their share toward boosting signal strength. (Signals from hundreds of different stations, some weak and some strong. may be present in a receiving antenna at any one time.)

The three tuning circuits in Fig. 17 are L_1 - C_1 , L_2 - C_2 and L_3 - C_3 . (In radio, we use the letter C to represent a condenser, and the letter L to represent a coil. When there is more than one coil or more than one condenser.



FIG. 17. Circuit arrangement of a typical t.r.f. receiver, with the two-stage r.f. amplifier circuit shown in simplified form. All d.c. circuits are completed through the power supply, which is not shown. All paths for signal currents are completed through condensers and other parts, which have been omitted here to simplify the diagram.



Courtesy General Electric Co.

Typical tuned radio frequency (t.r.f.) receiver. This set has five tubes, and uses circuits very similar to those which serve as examples in this lesson. Stations are tuned in with the large dial-knob, while the set is turned off or on and the volume is adjusted with the smaller knob.

numbers are used after the letters to distinguish between the similar parts.) Each tuning circuit contains a coil and a condenser. The condensers are shown separately in the diagram in Fig. 17, but in the actual receiver these three condensers are combined into a single unit like that shown at the upper right in the diagram. The dotted lines connecting together the three tuning condenser symbols indicate that these condensers are on the same shaft and tuned by the same knob.

When we tune in a program with the tuning knob, we are actually adjusting all three condensers at the same time so they will tune all of the tuning circuits automatically to the carrier frequency we desire to receive, and at the same time reject the undesired signals.

In the early days of radio, singlecontrol tuning was unknown. Each tuning circuit had its own separate tuning condenser. To tune in a station, it was necessary to adjust all of the tuning knobs, one at a time. Some sets had as many as five tuning condensers, each with its own knob, so it was quite a feat to tune in a station in those days.

Getting back to our own t.r.f. circuit in Fig. 17, the antenna signal current flowing through coil L induces a signal voltage in the first tuning circuit, L_1-C_1 . The tuning circuit transfers this weak voltage to the grid and eathode of the first r.f. tube (the C battery, not shown, completes the circuit from C- in the tuning circuit to C+ at the cathode).

As a result of the amplifying ability of the tube and its B voltage source, the modulated r.f. signal current flowing in the plate circuit of the first r.f. stage is much stronger than the antenna signal current. In Fig. 18 is a



FIG. 18. The first r.f. stage of Fig. 17 is redrawn here as a guide for you to follow in preparing the rough sketch called for in Question 9. Your sketch can be even rougher than this, just be sure it shows the circuit correctly.

rough sketch showing this first r.f. stage as it might be drawn by a radio man.

The second r.f. amplifier stage and the two remaining tuning circuits provide still more boosting of our signal, so that the signal voltage in tuning circuit L_s - C_s is strong enough to be handled by the detector.

For the present, it is sufficient to understand that the r.f. amplifier in a t.r.f. receiver accepts and steps up the voltage of a desired radio signal, and also keeps out undesired signals. By using a sufficient number of tuned stages, complete rejection of undesired stations can be obtained.



FIG. 19. Simplified detector circuit of a t.r.f. receiver.

The Detector. Now that the modulated r.f. signal has been built up sufficiently in strength, the r.f. carrier signal is no longer needed; in fact, we must get rid of the r.f. carrier now, so it will not interfere with the operation of the remaining stages. We separate the audio signal from its r.f. carrier with a vacuum tube operated in a special manner in the detector stage. A simplified version of one common type of detector circuit is shown in Fig. 19.

The modulated r.f. current which we feed into the detector is an alternating current, and its pattern on the screen of a cathode ray oscilloscope would be exactly the same as the modulated r.f. carrier pattern shown in Fig. 15C for the transmitting antenna.

The detector tube allows the input

signal voltage to act on the grid in one direction only. This has the effect of cutting off one entire half of this signal pattern, because it allows only half of each alternating current cycle to affect the plate current of the detector tube.

The signal pattern at the output of the detector tube would therefore be as shown in Fig. 20A. This pattern represents a d.c. voltage which is always acting in the same direction, but is varying in value from zero to a maximum at an r.f. rate (many hundred thousand times each second). The maximum value at each instant is determined by the audio signal.

The job of getting rid of the r.f. variations in the detector output signal is taken care of by two simple radio parts, condenser C and coil L.



FIG. 20. Here is the story of how the audio signal is separated from its carrier in the detector stage of a Lr.f. receiver, as presented by patterns on the screen of a cathode ray oscilloscope. The values (electrical characteristics) of these two parts are so chosen that condenser C lets r.f. signals pass through but blocks audio signals. Coil L, on the other hand, lets the audio signal pass through but blocks the r.f. signals. Together these two parts force r.f. signals to take the path through C from the plate to the cathode of the detector tube, and this is the end of the r.f. signal so far as our receiver is concerned.

The signal current which gets through coil L and also flows through the primary of the audio transformer in Fig. 19 is the *audio signal* we want. Its pattern is shown in Fig. 20B, and is simply the pattern of Fig. 20A with the vertical lines of the r.f. signal removed.

The audio transformer has more turns on its secondary than on its primary, and consequently it steps up the audio signal as well as transfers it to the next section, the audio amplifier. The pattern of our strengthened audio signal at the secondary of the audio transformer is shown in Fig. 20C. Notice that this strengthened signal swings much farther above and below the horizontal zero line than does the signal in Fig. 20B; this indicates the increased strength of the signal.

We thus secure from our detector circuit an audio signal like the original signal produced by the microphone. The one important fact to remember is that the detector in a radio receiver separates the audio signal from its r.f. carrier.

The Audio Amplifier. The audio output signal of the detector stage in a receiver is strong enough to operate headphones, but is usually too weak for loudspeaker operation.

We build up the strength of this audio signal in exactly the same way that we build up the strength of the microphone output signal in the transmitter, by means of audio amplifier stages using vacuum tubes.

A t.r.f. receiver may have one, two or even more a.f. amplifier stages; the more stages there are, the greater will be the audio signal power fed to the loudspeaker and the louder will be the maximum volume obtainable from the receiver. The last audio amplifier stage is often called the *output stage*.

The iron-core transformer which transfers the audio signal from the output stage to the loudspeaker is called the *output transformer*. The connections are shown in Fig. 21.



FIG. 21. Here is a simplified version of the very last stage in a radio receiver. It connects the power tube (the last tube in the audio amplifier) to the loudspeaker.

The output transformer is usually mounted directly on the loudspeaker, simply because this is a convenient position, but you will occasionally encounter sets which have the output transformer on the chassis.

For the present, all you need to know about the output transformer is that it supplies the correct signal voltage to the loudspeaker without too greatly affecting the operation of the output stage. Later, you will learn more about output transformers—why they are needed, why they are stepdown transformers, and how they can affect the tone quality and volume of a receiver.

The Loudspeaker. The newest type of loudspeaker, known as a permanent magnet (p.m.) dynamic loudspeaker, consists essentially of three parts: a powerful permanent magnet, a paper cone, and a voice coil. The voice coil is mounted at the center of the paper cone and is located between the poles of the permanent magnet, as indicated at the right in Fig. 21.

When audio frequency current is sent through the voice coil of a p.m. dynamic loudspeaker, a magnetic field is produced by the coil. This magnetic field reacts with the magnetic field of the permanent magnet, with the result that the voice coil moves either in or out. Since the diaphragm is attached to the coil, the diaphragm moves also.

The movements of the voice coil correspond exactly to the variations in the audio current passing through the coil, and consequently the diaphragm vibrates at audio frequencies and produces sound waves.

The sound waves produced by the loudspeaker are the same kind as those which acted on the microphone in the radio studio. They vary in the same manner and affect our ears in the same way, so we hear at the receiver the very same sounds which are in the studio. No wonder people say it is the magic of radio which transfers sounds thousands of miles in a split second!

A p.m. dynamic loudspeaker is similar to a dynamic microphone. In fact, in most intercommunication systems employed in factories and stores, a special loudspeaker of this type also serves as a microphone. The talklisten switch automatically connects the loudspeaker to the input of an audio amplifier for microphone operation, and to the output of the audio amplifier for loudspeaker operation.

The loudspeaker is the only part in a radio receiver which moves continually, and even its motion cannot be seen because it is so fast. In fact, all the signals in a radio system

travel faster than a bolt of lightning.

Even though you can't see radio receiver parts move like parts in a locomotive or a threshing machine, you will soon be able to visualize plenty of action in radio circuits. As you become more familiar with the tubes, transformers, resistors, condensers and other parts used in radio, you will find yourself thinking of electrical actions in these parts—actions more exciting and more rapid than in any man-made engine. Radio will become *alive*, and you will develop an *enthusiasm* which makes your radio job a pleasure rather than a chore.

Volume Control. The average radio receiver, when tuned to a local station, is capable of producing far more volume than is required by listeners in homes. Furthermore, people like to vary the loudness of a radio program to suit the occasion; they want much more volume for a news broadcast than for background music which accompanies the conversational chatter of a dinner party. For these reasons, every radio receiver has a device which serves as the volume control. It is adjusted by a knob or lever on the front panel of the receiver.

When you turn down the volume of a radio receiver by rotating the volume control knob counter-clockwise, you are simply reducing the strength (voltage) of the signal which is being transferred from one point to another in the receiver. When you advance the volume control all the way, for reception of a distant or foreign station, you are simply using the full signal-boosting ability of the receiver.

On-Off Switch. Another control which you will find in every radio receiver is an *on-off switch* or *power switch*. This switch turns off the receiver by disconnecting it from the

power line or other source of power. Assuming that our t.r.f. receiver operates from an a.c. power line, this switch would be located on the front panel of the receiver and would be connected into one of the leads of the receiver power cord.

Oftentimes the volume control knob also operates the on-off switch. Turning this knob a slight amount clockwise clicks on the switch, and further turning increases the volume.

Power Pack. An a.c. power line provides only an a.c. voltage, but the tube electrodes in a receiver require d.c. voltages like those obtainable from batteries. In order to change the a.c. line voltage into the various d.c. and lower a.c. voltage values required by the tubes, a special vacuum tube stage known as the *power pack* is needed in every receiver which operates from an a.c. power line.

Although receivers may have different types of power packs or may operate directly from batteries, you will find that all receivers—aircraft sets, auto radios, portable receivers, and others—have essentially the same signal circuits. This is indeed fortunate, for it means that you need to study only a limited number of signal circuits.

The vacuum tube used in a power pack is known as a *rectifier tube*. Its job is to convert the a.c. voltage into a *pulsating* d.c. voltage, which always acts in the same direction even though it rises and falls in value. Condensers and a coil acting together in a *filter circuit* smooth out the variations in this pulsating d.c. voltage, so as to give the *pure* (unvarying) d.c. voltage required by the tubes. The other important duty of the power pack is to provide the correct low a.c. voltage for each tube filament. In your very next lesson, you will study several ways in which this is done.

Avoiding Distortion of Sound. The sound as reproduced by the loudspeaker of a radio receiver will be identical with the original sound picked up by the microphone in the broadcasting studio only if our audio signal varies in exactly the same way at all points on the long path between the studio and the home. One improper adjustment or one defect anywhere along this path can destroy the faithfulness of reproduction of the sound, making repairs necessary. This unnaturalness of a radio program coming from a loudspeaker is called distortion.

Your Duty. As a radio serviceman, it would be your duty to maintain the receiver end of this radio broadcasting chain in perfect condition. As a communications expert, you would be intrusted with the care of the transmitter end.

The only part of the radio signal path which is completely out of the hands of radio men is the space between the transmitting and receiving antennas. Here nature produces electrical storms which cause static interference. In addition, the reflecting layers high in the stratosphere shift up or down from time to time and cause fading of radio signals. But even these natural enemies of radio are gradually being conquered by radio engineers, with improved circuits, improved equipment, and a greater knowledge of radio.

A Review

A brief review of the process of sending and receiving a radio broadcast will tie together in your mind the various steps involved in sending a sound radio program from the studio to your home.

Starting at the studio, the performers produce sound waves. These are picked up by the microphone and converted into an audio signal having the same characteristics as the sound waves. This audio signal travels through the microphone cable to the control room, where it is amplified many times by vacuum tube amplifier stages.

The amplified audio signal is then fed over telephone lines to the transmitter proper, which may be many miles from the control room and studio. The audio signal undergoes additional amplification at the transmitter before it is fed into the modulated stage. A radio frequency carrier voltage, produced by a crystal oscillator and amplified by various r.f. amplifier stages, is also fed into the modulated stage.

The modulated r.f. carrier signal of the modulated stage is amplified further by r.f. amplifier stages, and the resulting strong modulated r.f. carrier signal is then fed to the transmitting antenna. The electrons vibrating in the transmitting antenna under the influence of this modulated r.f. carrier current produce vibrating electric and magnetic fields which travel out through space as radio waves.



RADIO RECEIVER SERVICEMAN BROADCAST





Here are four important radio jobs for which this lesson gives essential background information.

World Radio History

At a receiver location, the free electrons in the receiving antenna wire are set into vibration at the frequency of the radio waves. The result is a very weak modulated r.f. carrier current in the receiving antenna. This is amplified thousands of times by the r.f. amplifier stages in the receiver, after which the signal is sent into the detector.

The detector stage separates the audio signal from its r.f. carrier, and feeds only the audio signal into the audio amplifier. The audio amplifier in turn delivers a strong audio frequency current to the loudspeaker; this current causes the loudspeaker; this current causes the loudspeaker diaphragm to push and pull against the surrounding air, creating sound waves which are a true reproduction of the sound produced by the performers in the broadcasting studio.

Looking Ahead

You have now obtained enough knowledge of radio to give you the necessary foundation for one of the most fascinating careers in the world —RADIO. Best of all, you didn't have to wade through fifteen or twenty lessons to get this information. You secured your foundation knowledge from just two lessons, because the N.R.I. Course gives you only what you really need.

You have now passed a real milestone, and you are certainly entitled to a satisfying feeling of accomplishment. At the same time, you have prepared yourself for the next milestone, which is a fascinating study of individual radio parts, circuits and meters. Here you will take up circuit details which_were purposely saved for this next lesson, and study many other interesting arrangements of radio parts which are sometimes used in place of the circuits already studied.

In the next lesson you get a wealth of practical information about meters, which are the most valuable tools of all for radio men. You will learn exactly how meters are used in real radio circuits, and how they tell you what the invisible electrons and radio signals are doing in your circuits. The ease with which you master this next lesson will be additional proof of how much you have learned in such a short time.

You will also learn about practical battery connections such as are used in portable receivers, and about different ways of changing voltages from one value to another.

With the confidence and "feel" of radio you have already developed, you will be anxious to start the next lesson as quickly as possible. Therefore, turn back to the first page now and take care of those last few steps in the Study Schedule, so you can start on the next lesson while the important facts of the first two lessons are still fresh in your mind.

Lesson Questions

Be sure to number your Answer Sheet 2FR-5.

Place your Student Number on every Answer Sheet.

Send in your answers for this lesson immediately after you finish them. Doing this insures that the graded answers will reach you while the subject matter is still fresh in your mind, and you will get the greatest possible benefit from our speedy personal grading service. Never hold up a set of lesson answers.

- 1. What is the approximate range of frequencies (the lowest and highest) which can be heard by human ears?
- 2. What is the important purpose of a microphone?
- 3. Write on your answer sheet the missing words which will correctly complete this statement:

- 4. Do the electrons in the voice coil of a dynamic microphone ever get into the transmitting antenna?
- 5. Name the two important signals which are combined in the modulated stage of a broadcast transmitter to produce the modulated r.f. carrier signal.
- 6. What two kinds of fields make up the radio waves (electromagnetic waves) which travel away from a transmitting antenna in all directions?
- 7. Is the modulated r.f. current in the receiving antenna STRONGER, WEAKER, or the SAME IN STRENGTH as the modulated r.f. current in the transmitting antenna? (Only one of these three possible answers is correct. Write down on your answer sheet the one which you think is correct.)
- 8. What are the two purposes of the r.f. amplifier section of a t.r.f. receiver?
- 9. Make a rough pencil or pen sketch of the simplified circuit given in Fig. 18 for the first r.f. stage of a t.r.f. receiver. (Any convenient size will do.)
- 10. What stage in a t.r.f. receiver separates the audio signal from its r.f. carrier signal?



YOU'VE GOT COURAGE

An eight-year-old boy was discussing arithmetic with a friend. Said Johnny: "Teacher's gonna start us on *subtraction* tomorrow; wish I could stay home." But Harry laughed outright at Johnny's fears, and said: "Aw, I've had that and it's easy once you get started; what I'm worrying about is *multiplication*!"

It is natural even for grown men to feel like these boys—to fear most the things about which they know the least.

Did you know that some of the world's best speakers are always afraid when they get up to make a speech before a strange audience? Their *courage* gets them started, and in no time at all their fear changes to a confident enthusiasm which makes their talk a big success.

You've got courage! Use your courage to overcome normal fears, to carry you into each new radio subject and carry you over each difficulty. In no time at all, you will be looking forward to new subjects with intense interest—you will be *eager* to tackle new problems. Remember that each *conquered* difficulty brings you one step closer to your goal of SUCCESS IN RADIO!

J. E. SMITH.



NATIONAL RADIO INSTITUTE ESTABLISHED 1914 WASHINGTON, D. C.



STUDY SCHEDULE NO. 3

By dividing your study into the steps given below, you can master this part of your N.R.I. Course quickly and thoroughly. Check off each step when you finish it.

How to Study Each Step. Unless otherwise instructed, first read the pages specified for the step at your usual reading speed. Next, reread them slowly enough so you understand every paragraph. Go over the material as often and as slowly as you find helpful, endeavoring always to *understand* as well as remember. Finish up with one quick reading to tie the ideas together in your mind, then write down roughly on scrap paper your answers to the questions specified for that step. After finishing the lesson, copy all your answers neatly and in order on an Answer Sheet.

□ 1. Meters Tell What Circuits Are Doing......Pages 1-2 Read this introductory section slowly just once. You will find out why meters are studied along with simple circuits. A few of the interesting ways in which you will use meters in your radio work are described.

- 3. How Voltage and Resistance Affect Current in Simple Circuits...Pages 6-10 Here is information which you will use in practically every radio job, so give it all you've got! Answer Lesson Questions 4 and 5.
- ☐ 4. Load Connections for Radio Circuits......Pages 10-13 You'll learn how loads can be connected either in series or in parallel to a voltage source, and you'll find out how a voltage-dropping resistor uses up some of the source voltage. Answer Lesson Questions 6, 7 and 8.
- □ 5. What Makes Meters Work.....Pages 14-21 Meters will mean a lot to you after you read this clear and interesting explanation of how the basic meter works, and how it is used with a few simple additional parts to give us all of the meters ordinarily used by radio men. Answer Lesson Question 9.
- 6. Voltage Sources for Radio Circuits......Pages 21-28 An interesting section dealing with storage batteries, dry cells, B batteries, and other voltage sources used in radio circuits. Answer Lesson Question 10, then read "Looking Ahead."
- 7. Mail Your Answers for Lesson 3FR-2 to N.R.I. for Grading
- 9. Start Studying the Next Lesson.

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World Radio History

SIMPLE RADIO CIRCUITS AND METERS

Meters Tell What Circuits Are Doing

RADIO receivers, transmitters, public address systems and photoelectric controls are all alike in the sense that each one is made up of various combinations of *simple circuits*. Once you are familiar with the few simple circuits which are used over and over again in radio apparatus, you will have a good practical start toward understanding the complete circuits of receivers and other equipment.

This third lesson of the N.R.I. Course is a particularly important part of your training, because it gives the practical facts you want to know about these simple circuits in order to become a capable and well-trained radio man.

In addition, this third lesson explains in detail how different kinds of meters are used to find out what is going on in simple circuits. Since meters are the radio man's most valuable tools, and since this is definitely a practical training course in radio, it is logical that you study meters along with circuits.

As you study this lesson, you are going to be pleasantly surprised at how much the previous two lessons are already helping you. You have really acquired a wealth of information about radio, and you will already be using this knowledge.

To illustrate the importance of knowing how to use meters, we will now consider a few of the ways in which meters are used by radio servicemen and radio operators on actual jobs. How a Serviceman Uses Meters. Let's look into the future for a few minutes now, and imagine that you are the owner of a prosperous radio servicing business on the main street of your home town. A customer brings in his table model radio receiver, say-



Serviceman using a d.c. voltmeter to check the plate voltages of tubes in a defective receiver. A few simple measurements with meters are often enough to locate a defective part.

ing: "It went completely dead last night, right in the middle of my favorite program. This never happened before!"

Taking the receiver to the test bench, you transfer the tubes one by one from the receiver to your tube tester, an instrument having a meter as its most important part. For every tube, the meter pointer swings to the green area marked GOOD TUBE.

1

The tubes are okay, so now you begin using your knowledge of radio fundamentals and servicing techniques. With the receiver turned on, you momentarily remove each tube in turn or touch its top cap. The resulting clicks or thumps in the loudspeaker for some but not all tubes tell you which stage of the receiver is at fault.

Now you reach for another instrument (a voltmeter) and measure the voltage at each electrode of the tube in this stage.

"Ah ha! No plate voltage. Something wrong here!", you exclaim in triumph. You have traced the trouble to one simple circuit, and a few quick measurements reveal the brokendown part which made the set go dead.

Your meters thus locate for you in a few minutes a part which is defective even though it looks perfectly good. After the meters have done their work, it takes just a few minutes longer to remove the defective part, put in a brand-new replacement part, then give the receiver a final complete check-up before handing it back to your customer and collecting your money.

Watching you at work in that imaginary radio shop of the future is a great deal like watching a doctor meet his daily assortment of patients. Just as the doctor uses thermometers, Xray machines, stethoscopes and other instruments to supplement his medical knowledge in analyzing and treating human ailments, so do you use meters of various types to supplement your radio circuit knowledge in locating the causes of receiver ailments.

How a Radio Operator Uses Meters. The scene changes. Imagine that you are a radio operator temporarily assigned to the evening shift at a powerful broadcasting station. From your control desk in the center of the transmitter room, you can see dozens of meters, arranged in rows on the metal panels of the transmitter. Each meter has its own particular story to tell you about one of the simple circuits in that transmitter.

Let's say it is 10 o'clock, with three more hours to keep the station on the air, when the pointer of one meter begins slowly creeping downward. The meter is telling you that one of the tubes is starting to fail.

You get a new tube and place it on your desk, ready for speedy replacement if necessary. But you don't put in the new tube now, because that would mean shutting off the transmitter. From experience, you know the tube will probably last quite a few hours yet anyway.

Fortunately, the tube does last. The announcer signs off for the night, you open up the master switches and disconnect all power from the transmitter, then you pull out the failing tube and put in the new tube.

Clearly, meters help you to keep a transmitter on the air. Without me-



ters, you wouldn't know anything was wrong until a tube actually failed, and then would lose many precious minutes in finding out which tube had failed.

In addition to helping you locate trouble, the meters in a transmitter tell you when each circuit is correctly adjusted so the transmitter and the transmitting antenna will be operating at maximum efficiency. Even the moni-

tor operator in the control room relies on a meter to aid in adjusting the volume control.



What Radio Men Measure

There are only three values which radio men ordinarily need to measure in a radio circuit: voltage, current and resistance.

You already have a general idea of what d.c. and a.c. voltages are like, and are also somewhat familiar with both alternating current and direct current, so we need only review these subjects briefly before taking up the new subject of electrical resistance.

A great many technicians think of voltages, currents and resistances as things which can be measured with meters. If you mention 45 volts to them, they immediately think of a meter with its pointer at 45 volts. Practical slants like this are emphasized throughout your N.R.I. Course, along with essential technical information.

Voltage

The electrical "pressure" which is capable of setting electrons into motion in a circuit is called *voltage*. Any device which produces a voltage is a *voltage source*. Storage batteries, dry cells, and generators are all voltage sources.

The stronger the voltage, the greater the effect it has on electrons. The term volt is used by radio and electrical men as a convenient unit for specifying the strength of a voltage source. For example, a 6-volt storage battery is stronger or higher in voltage than a 2-volt storage battery.

Voltages are measured with meters called *voltmeters*. Though scientists have elaborate ways of specifying how much a volt is, a radio man is satisfied to rely upon his voltmeter to tell how many volts a particular voltage source has.

D.C. Voltage. A voltage which

always moves electrons in the same direction is a *d.c. voltage*. It always has the same polarity. (One terminal is always + and the other always -).

The d.c. voltage sources most often used in radio are dry batteries, storage batteries, d.c. generators, and power packs. (A power pack converts a.c. voltage to d.c. voltage.)

A.C. Voltage. That type of voltage which reverses its polarity regularly, sending electrons through the circuit alternately in one direction and then in the opposite direction, is known as an *a.c. voltage*. The electrons do useful work, however, even though they just move back and forth many times each second without getting anywhere.

A.C. generators, ranging in size from the huge units in power plants down to small units driven by gas engines, are a.c. voltage sources. In addition, every radio signal source produces an a.c. voltage. Microphones, oscillator circuits, transformers and receiving antennas are thus all a.c. voltage sources.

How A.C. and D.C. Voltages Act Together. Here is a fact which is sometimes hard to believe: Both a.c. and d.c. voltages can exist in the same circuit.

When there are a.c. voltages in the same circuit with a d.c. voltage, each voltage will act independently just as if the others didn't exist. By using the proper meters, you can even measure these voltages independently.

Every amplifier stage is an example, for you have in the grid circuit a d.c. voltage (the C bias voltage), and an a.c. signal voltage provided by the signal source.

Pulsating D.C. Voltage. When a d.c. voltage is varying regularly in

value above and below a fixed value, without reversing its polarity, we have a *pulsating d.c. voltage*. It is simply a d.c. voltage combined with a smaller a.c. voltage, and these two voltages can be considered independently just as if they were separate voltages.

The most familiar example of a pulsating d.c. voltage is that produced by the rectifier tube in the power pack of a radio receiver. Remember, a pulsating d.c. voltage always has the same polarity, but changes in strength or value continuously.

Other Names for Voltage. You will sometimes find the terms *potential* and *electromotive force* (abbreviated *e.m.f.*) used in place of voltage. Potential means essentially the same as voltage, while e.m.f. is used only for voltages produced by true basic voltage sources like batteries or generators.

Current

The electron movement produced by a voltage is called a *current*. More generally speaking, any flow of electricity is a current.

The more electrons we have moving past a given point per second, the stronger is the current. The ampere (pronounced AM-peer) is the basic unit used for specifying the strength of a current. For instance, an automobile storage battery supplies about 5 amperes of current to an auto radio receiver, about 200 amperes to the starter, and about 7 amperes to each headlight bulb.

In radio work, we deal so much with currents smaller than one ampere that we frequently use a smaller unit called the *milliampere* to specify the strength of a current. A milliampere is onethousandth of one ampere, so 1,000 milliamperes is equal to one ampere.

Current values are measured with meters called ammeters or milliam-

meters. Ammeters indicate the current value in amperes. Milliammeters indicate the current value in milliamperes.

Since current-measuring meters give current values in amperes or thousandths of an ampere, there is no need for you to know how many electrons per second are involved in a current flow of 1 ampere. The actual figure is unimaginably large, and is presented here only as an oddity of science: It takes 6.3 million million electrons moving past a given point in one second to give a current of 1 ampere.

Direct Current. A current which consists of electrons flowing always in the same direction is a *direct current*. It is produced by a d.c. voltage.

With direct current, the direction in which electrons are moving is of importance only when we need to figure out the correct polarity for connections to batteries, meters or radio parts. When direction is important, we speak of *electron flow* and specify a direction; when direction does not matter, we just speak of *current*.

Both batteries and radio tubes can serve as "signposts" in circuits, just as is shown in Fig. 1. These parts tell us which way electrons are moving. Thus, electrons always move away from the *negative* battery terminal.



FIG. 1. Batteries and tubes are the signposts of radio circuits, for they tell you the directions of electron flow in d.c. circuits.

and move through the circuit towards the *positive* terminal of the battery.

In a tube circuit, the electron flow is always in such a direction that the electrons flow from the cathode to the plate in the tube.

Alternating Current. A current which consists of electrons which regularly reverse their direction of motion is an *alternating current*.

Pulsating Direct Current. A current which changes regularly in strength but still consists of electrons flowing always in the same direction is a *pulsating direct current*. It is produced by a pulsating d.c. voltage, which is really a combination of a d.c. voltage and an a.c. voltage.

Resistance

The opposition which a wire, radio part or circuit offers to the flow of electrons is called *resistance*.

Everything has resistance, but metals like copper and aluminum have very low resistance. These metals offer little opposition to electron flow, and are therefore considered to be good paths for electrons and good conductors of electricity. Copper is the metal most frequently used when resistance should be very low, such as in the connecting wires of radio circuits. The resistance of a copper connecting wire is so low that it is usually neglected by radio men, and the wire is assumed to have zero resistance.

Materials like carbon and nichrome have quite a high resistance, and are used in radio circuits when it is necessary to reduce the strength of a current to a desired value.

The Ohm. The unit used for expressing the *amount* of resistance is the *ohm* (pronounced like "ome" in "home," and named after the scientist George Simon Ohm). Radio men use

meters called *ohmmeters* to tell how many *ohms* of resistance a part has.

Resistors. When we want to reduce the amount of direct current flowing through a circuit, we connect into the circuit a radio part called a *resistor*. (*Resistance* is the actual opposition to electron flow, while a *resistor* is a part which has *resistance* or opposition to electron flow.)



FIG. 2. Here are some illustrations of typical radio resistors. The symbol used to represent resistors is shown in two forms—as it appears in printed diagrams, and as it would be roughly sketched by a radio man.

Two types of resistors are widely used in radio apparatus. Carbon-type resistors are made from a mixture of powdered carbon and a special cement. Wire-wound resistors are usually made from nichrome wire.

Examples of resistors you will work with are shown in Fig. 2, along with the symbol used on diagrams to represent a resistor. Practice drawing this zig-zag symbol which says to radio men "Here's a resistor."

The fifth lesson in your Course is devoted entirely to resistors, so we won't say much more about them now. For this lesson, all you need to remember is that a resistor is a radio part which *limits* current.

How Voltage and Resistance Affect Current in Simple Circuits

Every complete circuit has at least two parts, a *voltage source* and a *load*. The source and load can be connected together either by wires or radio parts.

The two things which affect the amount of *current* flowing in a complete circuit are the *voltage* of the source and the *resistance* of the load and other parts in the circuit. With the aid of meters, we can find out exactly what happens to the current when changes are made in voltage or resistance.

Continuing with our practice of talking in the radio man's language, we shall use throughout this lesson two terms which describe the two basic ways of connecting radio parts together in circuits. Here are the meanings of these terms:

In Series. When radio parts are connected together end to end, so that electrons flow through one part after another, we say that they are in series.

In Parallel. When radio parts are connected all to the same two terminals, so the electron flow divides among the parts, we say they are in parallel.

How Voltage Affects Current

Suppose you have a simple circuit consisting of a $1\frac{1}{2}$ -volt dry cell as voltage source and a 500-ohm resistor as load. You want to find out what happens to the *current* in the circuit when you increase the *voltage* by adding more dry cells *in series*.

Let us assume that for the current measurement you choose a milliammeter which reads from 0 to 10 milliamperes (abbreviated 0-10 ma.), and for the voltage measurement you choose a voltmeter which reads from 0 to 5 volts.

One Cell. You connect the volt-



FIG. 3. These simple circuits with meters show you how voltage affects the current in a circuit.
SOURCE VOLTAGE	CURRENT	
1/2 VOLTS	3 MILLIAMPERES	
3 VOLTS	6 MILLIAMPERES	
41/2 VOLTS	9 MILLIAMPERES	

FIG. 4A. This table gives the results of the measurements pictured in Fig. 3.

meter across the two terminals of the cell which serves as the voltage source, and connect the milliammeter in series with the load resistor, as shown in Fig. SA. The voltmeter reads $1\frac{1}{2}$ volts and the milliammeter reads 3 milliamperes (usually abbreviated as 3 ma.), so you record these values on a piece of paper, somewhat as shown in Fig. 4A.

Two Cells. The next question is —what will happen to the current if you double the voltage? To find out, you add another cell in series to your circuit, as shown in Fig. 3B, and read the meters again. The voltmeter reads 3 volts (twice the first voltage) and the milliammeter reads 6 ma. (twice the first current), so you record this set of values also in Fig. 4A.

Three Cells. Suppose you are still in an experimental mood, and add another dry cell to your circuit, as in Fig. 3C. Now the voltmeter reads $4\frac{1}{2}$ volts and the milliammeter reads 9 ma., so these values are recorded in Fig. 4A.

Conclusion. Now, what do these results mean? Well, by carefully studying and comparing the values in Fig. 4A, you could figure out for yourself the conclusion that current increases when you increase the voltage.

Graph Diagrams. It is not always easy to make correct conclusions just by studying a list of measured values, however. This is why radio men so often use a special diagram called a graph (pronounced GRAFF), which gives an easily-understood picture of the story hidden in a series of measured values. **Bar Graphs.** For example, we could take the values obtained in our little experiment and use them as the basis for a bar graph like that shown in Fig. 4B. One glance at this graph is almost enough to tell you that the current goes up when the voltage is increased.

Radio Graphs. The bar graph has



FIG. 4B. This graph presents the results of the Fig. 3 measurements in a manner more quickly understood than the table in Fig. 4A. You will often find bar graphs like this in newspapers, where they are used to show such things as the number of homes built each month of the year.



FIG. 4C. Here's the way a radio man would present the results of his measurements for the circuits of Fig. 3. This type of graph is easily drawn, and has the added advantage that it tells what the current will be for other values of voltage besides those measured.

one drawback for the radio man—it does not tell anything about values in between those actually measured. The radio man almost always uses a graph like that shown in Fig. 4C, because he can determine from this graph what the current will be for-any-value⁻ of voltage between 0 and 4.5 volts. Here is how this graph would be made up from the values in Fig. 4A.

Making a Graph. Starting with the first set of values in Fig. 4A, you locate 1.5 volts on the voltage scale at the bottom, and note the vertical (up and down) line which goes through this value. Next, you locate 3 on the current scale at the left and note the horizontal line which goes straight across the graph from this value. Move to the right along this line until you come to the vertical line going down to 1.5 volts, and at the crossing draw a dot or small circle, just as at a in Fig. 4C. This first point on your graph corresponds to the top of the shortest bar in Fig. 4B.

In the same way, you go up from 3 volts and to the right from 6 ma. until you reach point b on your graph and place a circle there. Finally, place a circle at c for 4.5 volts and 9 ma.

A bit of reasoning tells you that if the voltage is zero, no electrons will be flowing through the circuit, and the current will consequently be zero, so you place a small circle at 0 also.

Placing a ruler on your graph, you see that all four circles (0, a, b and c)are on a straight line. Therefore, as the final step in making this radio graph, you connect the four small circles together with a straight line.

Radio Curves. Whenever radio men connect together a series of points on a graph like this, they call the resulting line a *curve*. In fact, most of the graphs encountered in radio are

actually curves because the points are not on a straight line.

What a Graph Tells. A radio graph like that in Fig. 4C gives even more information than the original set of values. For instance, if you wanted to know what the current would be for a voltage of 2 volts in the circuit of Fig. 3, you would simply trace up from 2 volts till you hit the curve, then trace left to the current scale and read 4 ma. as the current. Try this yourself for 2.5 volts. If you get 5 ma. as the current for 2.5 volts, you have the right idea.

Remember This. The one important idea for you to get from this entire experiment is that when you increase the voltage, the current increases too. This basic rule works backwards just as well—when you decrease the voltage, the current goes down.

How Resistance Affects Current

Another Experiment. Now, let us take the very same circuit shown in Fig. SC, and see what happens to the current in the circuit when we increase the resistance.

One Meter is Enough. The only meter we need for this next experiment is a milliammeter, because we will leave the voltage source value fixed at $4\frac{1}{2}$ volts. Our new circuit arrangement is shown in Fig. 5A.

The milliammeter can be anywhere in the circuit, because the same amount of current flows through every part of a complete series circuit like this.

One Resistor. We start with one 500-ohm resistor, and get the same 9-ma. current as for the circuit of Fig. *3C*.

Two Resistors. How can we increase the resistance in this circuit? Well, the simplest way is by adding more 500-ohm resistors. If we con-



FIG. 5. These experimental circuits illustrate the basic fact that current GOES DOWN when you INCREASE the resistance in a circuit by adding resistors in series. If you start at circuit D and work back through C and B to A, these circuits will also illustrate the reverse condition, that current WILL INCREASE when you REDUCE the resistance in a complete circuit.

nect them in series, their resistance values will add.

Thus, two resistors in series as in Fig. δB give us a total circuit resistance of 500 + 500, or 1000 ohms. We now have twice as much resistance in the circuit as before, and the milliammeter now reads $4\frac{1}{2}$ ma. This, incidentally, is exactly one-half of the reading we obtained in Fig. δA .

Three Resistors. We continue increasing the resistance in the circuit, and make another measurement with three 500-ohm resistors in series to give a total of 1500 ohms. The milliammeter reads 3 ma. now, which is less than for either of the two previous measurements.

Nine Resistors. Finally, we take nine 500-ohm resistors in all, and con-



nect them in series as shown in Fig. 5D. Adding the values of these resistors together gives us now a total circuit resistance of 4500 ohms. With this resistance, the milliammeter reads 1 ma., which is still lower than before.

The Reverse Situation. If we start with high resistance in our circuit, as in Fig. δD , we will have an initial current of 1 ma. Now if we decrease the amount of resistance in our complete circuit by removing six resistors, the current will *increase* to 3 ma., as in Fig. δC . Decreasing the resistance still more, as in Figs. δB and δA , makes the current increase more yet.

Conclusion. The important facts to remember are that circuit current *decreases* when we *increase* the circuit resistance, and circuit current *increases* when we *decrease* the circuit resistance.

A. C. Measurements

Measurements in a.c. circuits are made in essentially the same way as d.c. measurements. Special a.c. meters must be used for a.c. measurements, and polarity of connections does not matter, but a.c. meters measure voltage and current just like d.c. meters. You're interested in actual valuesvolts, amperes or milliamperes—and that's what a.c. meters show.

The amount of current flowing in an a.c. circuit depends upon the source voltage and the circuit resistance, just as it_did for d.c. circuits.



Just two terminals that's all you'll find at the back of practically any radio meter. On d.c. meters, the positive terminal will usually be marked +. The other terminal, not marked, is negative (-).

Load Connections for Radio Circuits

Radio Loads. In radio, loads are grouped together quite frequently. For example, a radio receiver may have from 4 to 10 or even more tubes, with each tube filament acting as a load and hence requiring a voltage. Instead of using a separate filament voltage source for each tube, we can connect the filaments together either in series or in parallel and use one common



FILAMENTS IN SERIES (UNIVERSAL A.C.-D.C. SET)

FIG. 6. Practical example of loads connected in series. This example is taken from an actual receiver, and the type numbers of the tubes used in



the receiver are given above the tube symbols. Notice how the first two numerals in the type number of a tube tell its filament voltage. This holds true only for newer tubes, however. voltage source (or two at the most) for all of the filaments.

In a series connection of loads, the same current flows through all of the loads, one after another. In a parallel connection of loads, a different current flows through each load. With this distinction between series and parallel connections in mind, let us consider examples of these connections in radio receivers.

Loads in Series. When loads are connected together *in series*, the same current flows through all the loads. The total available voltage is divided among the loads, however, in proportion to the resistance of each load. The *sum* of the individual load voltages is then equal to the total voltage available for all the loads. Let us see how this applies to such well-known loads as tube filaments.

Filaments in Series. The filaments of radio tubes can be connected together *in series* if all of the filaments have the same current rating and the available voltage is high enough to take care of all the required filament voltages added together. The filament circuit arrangement shown in Fig. 6 for a 5-tube universal a.c.-d.c. receiver is a practical example of this. All the tubes have the same filament current ratings, and the power line provides a voltage which is just about equal to the sum of the required filament voltages. (A tube will operate satisfactorily even if the filament voltage is a bit higher or lower than the rated value.)

Loads in Parallel. When loads are connected together in parallel, all loads get the same voltage. The current flowing through each of the loads will depend upon the resistance of that load. Again we have tube filaments as examples, so let us consider the conditions under which these familiar loads can be connected in parallel.

Filaments in Parallel. The chief requirement for connecting the filaments of radio tubes together in parallel is that all of the tubes must have the same filament voltage rating. The available source must, of course, provide this voltage value, and in addition must be able to furnish a current



TTO A.C. POWER FILAMENTS IN PARALLEL

FIG. 7. Practical example of loads connected in parallel. The tube numbers are those of an actual receiver using this connection, so you are already studying real receiver circuits.

equal to the sum of all the rated filament currents.

The arrangement in Fig. 7 is a practical illustration of filaments connected in parallel. The five tubes each require a filament voltage of 6.3 volts. On the power transformer is a secondary winding which provides 6.3 volts,



Courtesy Lonith Radio Vorp.

Before each man on this final inspection line of a huge radio factory are meters which tell whether or not each receiver performs as it should. No matter what radio factory you visit, you will see hundreds of meters in use, telling radio men what is going on in radio circuits.

and all five filaments are connected in parallel to this winding.

Voltage-Dropping Resistors

Too Much Voltage. Very often in radio, we have a voltage source which provides a certain definite voltage value, and we have a load which requires a lower voltage value. In other words, the load requires only a portion of the source voltage, and there are no other loads which can use up the remainder of the voltage.

Getting Rid of Voltage. We take care of a surplus-voltage condition in radio work simply by inserting in the circuit a *resistor* which will use the undesired part of the source voltage. This resistor uses up surplus voltage in much the same way that automobile brakes use up part of the forward-



NOTE: When a radio man encounters a number like 68.9, he calls it 69 because that is sufficiently accurate for practical radio purposes.

acting force of an automobile when we want to slow down.

Here's an Example. A practical example would be a universal a.c.-d.c. receiver having a tube line-up like that shown in Fig. 8. Here the line voltage provides somewhere between 105 and 125 volts, but the tube filament voltage requirements only add up to about 69 volts. Assuming for convenience a line voltage of 115 volts, this leaves 46 volts to be used up somewhere in the circuit. We use up this unwanted 46 volts with a resistor of the correct value, inserted in series with the filaments just as shown in Fig. 8. We can insert this resistor anywhere in the circuit, because the same current flows through all parts of this series circuit.

What's Its Name? When a resistor is used in the manner just described, the resistor is called a *voltage-dropping resistor*. In effect, it *drops* the source voltage to the correct value for the load.

Voltage Rule. This example illustrates one of the basic rules in radio, that in any one circuit you have to use up all of the source voltage. If you don't need it all in the load, you have to insert resistance. Otherwise there'll be too much voltage for the load and something will burn out.

Voltage Drops. A voltage drop is simply some portion of a source voltage, appearing across a load, resistor or other radio part. Whenever electrons flow through some part, a voltage drop exists across that part.

When there are no voltage-dropping resistors in a circuit and there is only one load, the entire source voltage will act on the load. The voltage drop across the load will in this case be ex-



FIG. 9. This is how you would measure the power line voltage in a four-tube universal a.c.-d c. receiver.

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actly the same as the source voltage. An example of this is an ordinary electric lamp bulb. It is connected directly across the 115-volt voltage source, so the voltage drop across the lamp is likewise 115 volts.

Experimental Proof. Imagine that you are going to prove for yourself the basic rule that all of the voltage drops across loads and resistors in a circuit do add up exactly to the source voltage. You decide to use a universal a.c.-d.c. receiver having the tube arrangement shown in Fig. 9.

First of all, you measure the line voltage by connecting your voltmeter exactly as indicated on the diagram. Suppose you find it to be 120 volts. Now, if the basic rule is to be proved, you must show that the voltage drops across the tube filaments and across the resistor add up to this value.

You measure these individual volt-

MEASURE BETWEEN	VOLTAGE IS	COMMENTS
A and B	I2 VOLTS	Filoment voltage of Type 12Z3 tube
B and C	25 VOLTS	Filoment voltage of Type 43 tube
Cand D	6 VOLTS	Filoment voltage of Type 6C6 tube
Dand E	6 VOLTS	Filoment voltage of Type 6D6 tube
E and F	71 VOLTS	Voltage drop across resistor
TOTAL	- I20 VOLTS	This is equal to the Line Voltage

FIG. 10. Here are the results you would obtain if you made measurements on the receiver represented in Fig. 9 to prove that the sum of the voltage drops equals the line voltage.

ages one at a time and jot down their values in the manner shown in Fig. 10, then add up the five measured values. Yes, they do add to 120 volts, so you have proved conclusively that the resistor voltage drops in this series circuit add up to the source voltage.

Terminals. Whenever two termi-

nals are connected together by a wire, as are the two terminals marked B in Fig. 9, a connection can be made to either terminal for voltage-measuring purposes, because there is no noticeable voltage drop in the short length of wire which joins these terminals.

Adding Voltage Drops. The voltage across two or more parts is simply the sum of the individual voltage drops. Thus, if you measure between points A and C in Fig. 9, you will be measuring the sum of the voltage drops across the filaments of the types 12Z3 and 43 tubes. The data in Fig. 10 indicates that these voltages are 12 volts and 25 volts respectively, and consequently you would measure 37 volts between A and C.

The voltage drops across the resistors in a series circuit must add up to a value exactly equal to the source voltage. This is an important fact for you to remember.

Polarity of Voltage Drops. It is occasionally necessary to measure the voltage drop across a load with a voltmeter. Whenever we work with d.c. voltages, the + terminal of the voltmeter must go to the + terminal of the voltage being measured, and hence we may sometimes have to figure out the polarity of the voltage drop across a resistor.

When you know the direction in which electrons are flowing through a resistor or load, the polarity of the voltage drop across the part can be determined by the following simple rule:

When electrons are flowing through a resistor, the polarity of the resistor terminals WITH RESPECT TO EACH OTHER is such that electrons ENTER the - terminal and LEAVE the + terminal.

What Makes Meters Work

The Basic Meter. The d.c. milliammeter is the heart of practically all meters used in ordinary radio work. Once you have a general idea of how this basic meter is constructed and how it works, you can very quickly learn how the d.c. milliammeter movement is used in d.c. voltmeters, ohmmeters, a.c. milliammeters, and a.c. voltmeters.

Looking at a Meter. From the outside, about all you can see of any meter are the terminals at the back, the glass window at the front through which can be seen the pointer which moves over the meter *scale*, and possibly also a part of the meter coil. Inside the case is where the differences between meters occur.

It Works Like This. The basic idea underlying a d.c. milliammeter can be expressed very simply. Suppose we have a permanent magnet of the modified horseshoe shape shown in Fig. 11A, with a much smaller permanent magnet pivoted between the poles of the large magnet.

Remembering the law of magnetism which says that like poles repel, you can see that when the small magnet is in the position shown in Fig. 11A, there will be a repelling action which rotates the small magnet in the direction indicated by the curved arrow.

D. C. Milliammeter

In a practical d.c. milliammeter, we have exactly the same situation except that our imaginary small pivoted magnet is replaced by a pivoted meter coil or electromagnet like that shown in Fig. 11B, through which we send the current to be measured.

The meter coil is pivoted on tiny jewel bearings so it will turn easily. A pointer attached to the top of the coil sweeps over the meter scale. Delicate spiral springs hold the coil in a position which places the pointer at zero on the scale when no current is flowing.

How a Meter Coil Acts. When current is sent through the meter coil, the coil becomes an electromagnet. For one direction of electron flow through the coil, this electromagnet might have the polarity indicated in Fig. 11B, which is the same as the polarity of the small pivoted magnet in Fig. 11A. The coil will react in the same way as the pivoted magnet; that is, the coil will rotate clockwise, and make the pointer move to the right. The greater the current sent through



FIG. 11. How a d.c. milliammeter works, and what the inside of an actual meter looks like.

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a meter coil, the stronger is the magnetic effect and the farther the coil rotates against the retarding action of the spiral springs. The meter coil always moves to a position at which the electromagnetic force which turns the coil is equal to the retarding force of the spring.

The divisions on the meter scale are spaced and numbered so the pointer will indicate the amount of current flowing through the meter coil.

We Take a Meter Apart. Figure 11C shows what you would see if you removed the case of a typical d.c. milliammeter. It should be pointed out, however, that radio men rarely if ever remove the case from a meter. Meters are even more delicate instruments than watches, and are easily damaged once the case is removed. Meters should be sent to the factory or to a meter repair expert when damaged.

Uses for D.C. Milliammeters. D.C. milliammeters are used by radio men to measure small values of direct current. Usually, milliammeters have coils which are designed to give a maximum (full-scale) reading for a current of 1 milliampere. Quite often, however, you will encounter milliammeters which are even more sensitive than this, and give full-scale readings on only a fraction of a milliampere.

Warning! Any current appreciably larger than the full-scale value of a milliammeter may damage the instrument. Excessively large currents will overheat the coil enough to melt the wire, and radio men then say that the meter is *burned out*.

Remember that whenever you use a milliammeter, make sure the current to be measured is within the safe current range of the meter.

Second Warning! Another equally important rule when using milliamneters is that a milliammeter must always be placed *in series* with the circuit. In other words, you must actually cut into a circuit in order to connect a milliammeter properly. NEVER connect a milliammeter across a voltage source or across any part having a voltage drop, because this is almost certain to burn out the meter.

Higher Meter Ranges. Milliammeters and ammeters of all sizes employ essentially the same basic meter



FIG. 12. This milliammeter, which reads currents up to 10 ma., is nothing more than a basic 0 to 1 ma. d.c. milliammeter with a shunt resistor.

construction illustrated in Fig. 11. Higher ranges are secured simply by adding to this basic meter a resistor which provides a short-cut or shunt path for some of the current being measured. The resistors used for this purpose are known as *shunts*, and are usually built right into the meter.

Figure 12 illustrates how a shunt is connected to a basic meter. By using various sizes of shunts with a basic 0 to 1 ma. d.c. milliammeter, you get milliammeters which can read from 0 to any higher maximum value such as 5 ma., 30 ma., 100 ma., or 500 ma. By using shunts still larger in dimensions, so that they have lower resistance and take a still greater portion of the current, you get ammeters which can read from 0 to any desired maximum current value in amperes.



FIG. 13. The notes above this typical meter scale tell you what a radio man would read at various pointer positions. Figure out for yourself the value of each unmarked division line on the scale. Thus, the first line to the right of 1 is 1.1, the second is 1.2; the third (just past 1.25) is 1.3; the fourth (one line to the left of 1.5) is 1.4; etc.

Polarity for Meters. Radio men usually try to get the correct polarity the first time when connecting a d.c. milliammeter, because this saves quite a bit of time.

The correct polarity of connections for a milliammeter is such that the + terminal of the milliammeter goes to the + terminal of the voltage source, directly or through radio parts. The - terminal of the milliammeter then goes to the - terminal of the voltage source directly or through radio parts.

Just remember: Plus to plus, and minus to minus. This rule applies to d.c. ammeters, d.c. milliammeters and d.c. voltmeters.

If there is any doubt as to the correctness of a milliammeter connection, the radio man turns on the power for just a brief instant after making the connection and watches the meter pointer. If it starts to move backward (to the left of zero), he knows he must reverse the meter connections.

Reading a Meter Scale

A careful study of the enlarged meter scale in Fig. 13 will give you a good start toward reading meters. Once you understand what various pointer positions on this scale mean, you can apply your knowledge to practically any other meter scale which you may encounter in radio work.

Marked Divisions. When the pointer stops on a marked division such as .5, 1, 1.5, 2, 2.5 or 3, you simply read the printed number for that division line.

Unmarked Divisions. When the pointer stops on a small unmarked division, note the values of the marked divisions on either side, then figure out the value of the mark underneath the pointer just as you would figure out the value of one of the marks between 2 inches and 3 inches on a one-foot ruler.

In Between Divisions. If the pointer stops halfway between two unmarked divisions, determine the value of each of these unmarked scale divisions, then read a value halfway between them.

Estimating is O.K. It is entirely adequate in radio work to estimate meter readings roughly, when the pointer does not fall directly on a division line. Study the diagram in Fig. 13 until you are sure you could give the value corresponding to each division line on this scale. The notes above the diagram will help you to determine what these values are and how they would be read by a radio man.

D.C. Voltmeter

A d.c. voltmeter is nothing more than an ordinary d.c. milliammeter with a resistor placed in series with the meter coil.

This resistor uses up or drops a definite portion of the voltage being measured, and thus limits the current which can flow through the milliammeter.

Multiplier Resistor. The higher the resistance in ohms of a voltmeter resistor, the higher is the voltage range of the meter. The resistor in a voltmeter is usually called a *multiplier resistor*, for in a sense this resistor *multiplies* the useful range of the meter.

By using a printed scale made especially for use with the resistor in the meter, the pointer will correctly indicate the d.c. voltage which is being measured.

Uses for Voltmeters. Radio men use d.c. voltmeters to measure the d.c. voltages which exist between various tube electrodes and the chassis of a receiver, for these voltages are an important clue to trouble. In addition, d.c. voltmeters are used to check battery voltages, measure d.c. voltage drops across resistors and other radio parts, and for all other voltage measurements in direct current circuits.

Polarity Is No Problem. Although d.c. voltmeters must be connected with correct polarity before an accurate reading can be made, radio men ordinarily do not bother to figure out what the polarity is in case it is not marked. They connect one voltmeter lead to the circuit, then tap the other voltmeter lead momentarily on the other circuit terminal while watching the meter pointer. If the meter pointer starts to move backward (to the left of zero on the scale), they disconnect the meter and reverse its connections.

One Warning. The important thing to remember when using a d.c. voltmeter is that the voltmeter range must be *higher* than the voltage being measured.

Whenever in doubt as to what a voltage may be, start with the highestrange voltmeter you have, and get a rough idea of what the voltage is before selecting a lower-range meter which gives a more accurate reading.

D.C. meters are built ruggedly



Courtesy Western Electric Co.

Meters tell this radio operator what's going on in each circuit, and thus help him to locate the defective tube or part speedily when trouble occurs. This transmitter is at the Springfield, Illinois, headquarters of the Illinois State Police radio system. It operates on a Federallyassigned frequency of 1,610,000 cycles (1610 kilocycles), which is only slightly higher than that of a broadcast station. enough to stand considerable voltage overload. However, a voltage which is much too high for a meter may slam the pointer to the extreme right and bend it, or may even cause the meter coil to burn out.

Ohmmeter

An ohmmeter is nothing more than a d.c. milliammeter with a small battery and a resistor mounted inside the meter and connected in series with the coil of the meter, essentially as shown in Fig. 14.

Ohmmeter Uses. An ohmmeter is used by radio men to measure the resistance values of resistors or other radio parts.

Scale is Backward. The commonest type of ohmmeter has a scale which reads from right to left, which is backward when compared to the other meters we have studied. In other words, zero ohms on the scale is at the extreme right, and the highest resistance value in ohms is at the left. A typical ohmmeter scale is shown on the meter in Fig. 14.

Ohmmeter Zero Adjustment. The resistor inside an ohmmeter is variable in value, and is adjusted by means of a knob or screwdriver adjustment so as to bring the pointer to 0 on the scale when the test probes are held together to give zero resist-Since the ohmmeter battery ance. voltage drops with use, the resistor must be adjusted from time to time. This is known as an ohmmeter zero adjustment, and takes only a few seconds to do.

How an Ohmmeter Works. When a resistor is placed between the test probes of the ohmmeter, this resistor reduces the amount of current which flows through the meter, and consequently the meter pointer takes a position somewhere on the scale. (Remember that the meter itself is still



FIG. 14. Put a resistor and battery of the correct values in series with the coil of a d.c. milliammeter, and you have an ohmmeter for use in measuring resistance values. Polarity does not ordinarily matter when using an ohmmeter.

our basic d.c. milliammeter, with its pointer dropping to the extreme left when current is zero.)

The higher the value in ohms of the resistor being measured, the greater will be the total resistance in the ohmmeter circuit, the lower will be the current flowing through the circuit, and the closer to the left end of the scale will be the meter pointer. This is why resistance values on the scale of an ohmmeter must be zero at the extreme right and must increase gradually toward the left.

What Infinity Means. When the

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test probes are apart, there is an open circuit. This is the same as having an extremely high resistance between the ohmmeter terminals. Under this condition, no current flows through the meter, and the pointer drops to the extreme left. We say that the resistance is then *infinitely* high, which means that it is the highest value it is possible to think of—many million million ohms. Sometimes we use the symbol ∞ , called *infinity*, at the extreme left on the meter scale to designate an infinite resistance.

Warning. An ohmmeter must never be used in a circuit in which a voltage exists. Apparatus should always be disconnected from its power source before the ohmmeter is used. The ohmmeter has its own voltage source, and sends its own current through the part to be measured. Any additional voltage in the circuit would send extra current through the ohmmeter, possibly ruining the meter coil.

A.C. Milliammeter

The a.c. milliammeter used most often in radio work is nothing more than an ordinary d.c. milliammeter with one additional part called a *rectifier*. This rectifier is a tiny unit mounted inside the meter case, and has the job of changing the alternating current being measured into a direct current. The d.c. milliammeter responds to this direct current, but its scale is made to read in terms of the alternating current being measured.

Shunts. The ranges of a.c. milliammeters are increased by using shunts, just as with d.c. milliammeters. These shunts are usually built right into the instrument.

What the Meter Reads. You will recall that an alternating current is one which varies continually from a maximum value flowing in one direction to a maximum value flowing in the opposite direction. What is the a.c. milliammeter going to read in a case like this?

Here is the answer—the meter reads the a.c. current value which will produce the same amount of heat as the corresponding direct current value. Thus, an alternating current of one ampere will produce *the same amount of heat* as a direct current of one ampere.

A.C. Voltmeter

If you take an a.c. milliammeter and place a resistor in series with everything else inside the instrument, you have an *a.c. voltmeter*. The value of the resistor determines the range of the voltmeter.

Uses. A.C. voltmeters are used by radio men for checking the a.c. power line voltage, the filament voltages of tubes, and the various voltages across the secondary windings of the power transformer in a receiver or other piece of radio equipment. A.C. voltmeters are sometimes also used for measuring audio signal voltages.

What the Meter Reads. You will recall that an a.c. voltage is one which varies continually from a maximum value acting in one direction to a maximum value acting in the opposite direction. What is the a.c. meter going to read in this case?

The answer here is the same as for current—the meter reads the a.c. voltage value which will do just as much useful work as the corresponding d.c. voltage value. We call this a.c. voltmeter reading the *effective value* of the a.c. voltage. It so happens that the effective value of an a.c. voltage is equal to about seven-tenths of the maximum a.c. voltage acting in either direction.

Effective Value of Voltage. All you need to remember is that a.c. volt-



FIG. 15. Multimeter section of a typical professional all-purpose tester used by Radiotricians.

meters measure the *effective* value of an a.c. voltage. The a.c. voltages you will deal with are effective values unless otherwise specified. Thus, if you have an ordinary a.c. power line coming to your home, its 115-volt value is an effective value, and the voltage actually reaches about 165 volts during the peaks of each cycle.

Multimeters

Since every one of the meters ordinarily used in radio work consist of a d.c. milliammeter used alone or with resistors, rectifiers, shunts or batteries, it is only logical to make a single d.c. milliammeter serve for all purposes. This can be done by means of a switching arrangement which automatically connects the correct parts to the meter for a particular measure-The resulting many-purpose ment. instrument is known under various names, such as a multimeter, an allpurpose tester, a multi-tester, or a volt-ohm-milliammeter.

All-Purpose Tester. A typical ex-

ample of an all-purpose tester used by radio men is shown in Fig. 15. This one instrument is equal to twenty different separate meters, including five d.c. voltmeters with different ranges, five different a.c. voltmeters, three different d.c. milliammeters, one d.c. ammeter, four different ohnmeters and two different audio output voltage meters. This tester is thus capable of making practically all of the measurements required by radio men.

To prepare this all-purpose tester for use as a particular type of meter,





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all you need do is plug the test leads into the correct jacks on the panel, then set the two switches to the correct positions for the type of meter and range desired. Each jack and switch is clearly marked, so this can be done without even referring to an instruction book.

The meter in this tester has four scales. One serves for all d.c. readings, another for all a.c. readings, and the remaining two are used for the ohmmeter ranges. The two ohmmeter zero adjustments are of the screwdriver type, and are located at the lower left of the meter.

When the switch at the left of the meter is in its a.c. position, a rectifier is automatically inserted in the meter circuit for a.c. measurements. When this switch is in the d.c. position, the rectifier is disconnected.

Although for convenience and clarity we usually show individual meters in lessons, remember that a radio man would normally use a multimeter or all-purpose tester for a measurement.

Summary of Meters

The three important rules you should keep in mind whenever making meter measurements are:

1. For current measurements, always break into a circuit, so that the meter is in series with the circuit. This rule is illustrated in Fig. 16A. The current measurement can be made at any one of the four meter positions shown in this diagram.

2. For voltage measurements, connect to the points across which is the voltage to be measured. This rule is illustrated in Fig. 16B.

S. For ohmmeter measurements, connect across the resistance to be measured, but be sure to disconnect all batteries or other voltage sources, so no voltage exists in the external circuit. This rule is illustrated in Fig. 16C.

Voltage Sources for Radio Circuits

Since every radio circuit must have a voltage source of some kind, a general knowledge of the different kinds of voltage sources is highly desirable.

Many of the voltage sources used in radio are quite simple and familiar to most people. You may already know quite a bit about dry batteries and storage batteries, but even if you don't, you will undoubtedly find them easy to study.

The sections on d.c. and a.c. generators will particularly help you with radio work, because they give you a clearer understanding of what a.c. and d.c. voltages are like.

Batteries

Uses. Dry batteries are used in battery-operated farm radios, in port-

able radio receivers, in the portable two-way radiophone sets of the Army and the U. S. Forest Service, in radio test instruments and in many other types of radio equipment. Storage batteries are the main voltage sources for radio equipment in airplanes, boats, automobiles, and other large mobile units. Thus, batteries are still used enough in the field of radio to make a general knowledge of their construction and connections desirable.

Two Types. Two distinct types of units or cells are used in batteries: 1. Primary cells, which are used in dry batteries, and cannot be recharged when exhausted; 2. Secondary cells, which are used in storage batteries, and can be recharged.



FIG. 17. Construction of a standard dry cell.

Both primary and secondary cells produce a d.c. voltage by chemical action, and supply direct current. Scientists would say that both types of cells convert chemical energy into electrical energy.

Dry Cells

The dry cells generally used in portable radio receivers and in flashlights are perhaps the best known examples of primary cells. The chemical materials in a dry cell are gradually used up when current is drawn from the cell. When the supply of the chemical material is exhausted, the useful life of the cell is ended, and the cell must be replaced.

Construction. The construction of an ordinary dry cell is shown in Fig. 17. Flashlight cells are made in much the same way, but are smaller. In a flashlight cell, the negative connection is made *directly* to the exposed zinc case at the bottom of the cell, instead of to a top terminal attached to this case.

Voltage Produced. All ordinary dry cells deliver the same voltage of slightly over $1\frac{1}{2}$ volts (1.5 volts) when new. This voltage drops during use, and the cell must be replaced when its voltage becomes too low for its intended purpose.

Current Furnished. The larger the physical dimensions of a dry cell, the greater the amount of current it can deliver, or the longer it can deliver some definite value of current.

Dry Cells in Series. When a voltage higher than $1\frac{1}{2}$ volts is required, dry cells are connected together in series until the desired voltage is obtained. In this series connection, the — terminal of one cell goes to the + terminal of the next cell. You would connect cells together in series whenever you needed a voltage equal to the sum of the individual cell voltages.

For example, when a voltage of 45



FIG. 18. How 30 dry cells are connected together in series to make a 45-volt B battery.

volts is desired, thirty individual dry cells are connected together in series in the manner shown in Fig. 18. This arrangement gives us a 45-volt B battery, of which a cut-away view is shown in Fig. 19.

Dry Cells in Parallel. When you require more current than it is desirable to draw from a single cell, you can connect additional cells *in parallel*. Each cell then furnishes its share of current to the load, so you increase the current-furnishing capacity without affecting the voltage. Examples of parallel connections are shown in Fig. 20.

Remember This. When you want



FIG, 19. Part of this 45-volt radio B battery has been cut away to show some of the individual dry cells.

to get the greatest possible voltage from a group of dry cells, connect them together *in series*. When you want to get the greatest possible current or the longest life from a group of dry cells, without any more voltage than is provided by a single cell, connect them together *in parallel*.

Series-Parallel Connections. When both higher voltage and higher current-delivering ability are desired, a series-parallel connection can be used. This involves connecting one group of cells together in series until it has the desired voltage, then connecting



FIG. 20. Parallel connections of dry cells. Remember that when cells are connected together in parallel, the voltage is that of one cell, but the current-furnishing ability is increased.

this group of cells in parallel with two or more similar groups of cells until the desired current-handling ability is obtained. Examples are shown in Fig. 21.

Storage Batteries

The storage battery used in automobiles is the best-known example of a battery using rechargeable secondary cells.

Construction. A storage cell has two groups of plates, one group attached to the positive terminal and the other group attached to the negative terminal. These positive and negative groups of plates are made of lead (the metal, pronounced *LED*), and fit together alternately in the manner shown in Fig. 22A. Each plate has holes filled with the active chemical material of the cell, which is in paste form.

Between the plates are sheets of insulating material called separators, made either from porous wood or perforated rubber (Fig. 22B). Separators prevent the plates from touching each other and short-circuiting the cell.



FIG. 21. Examples of series-parallel connections of dry cells.

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Hydrometer Test. The plates of a storage cell are suspended in a container filled with a solution of sulphuric acid and pure water. The amount of sulphuric acid in the solution is an indication of how much the cell is charged (how much energy it contains in chemical form). The device shown in Fig. 23, called a hydrometer (pronounced hie-DROM-ehtur), is used to test the charge of a storage battery.

Recharging. The chemical materials change gradually to inactive forms when electrons flow *out of* the negative terminal of a storage battery during use. A storage battery is recharged by properly connecting it to a d.c. voltage source having a higher voltage than the storage battery. Charging is completed when the resulting reversed current has changed all of the chemical materials back to their original form.

Voltage Produced. The voltage of a storage cell is about 2 volts, and changes very little with use. For automotive purposes, three storage cells are connected in series, so that FIG. 22B. Cut-away view showing the construction of a typical 6-volt storage battery, which contains three storage cells connected in series.

their voltages add together to give the required 6 volts.

Storage batteries used for radio equipment in boats and airplanes often have six cells in series, to give 12 volts. The new tiny spill-proof storage batteries designed for portable radio receivers have only one cell, and hence deliver only 2 volts.

Generators

The majority of radio receivers in use today operate from wall outlets



FIG. 23. A hydrometer like this is used to test the condition of each cell in a storage battery. A fully charged battery will give a reading of about 1.280, and a discharged battery will give a reading of about 1.120.

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which furnish either a.c. or d.c. voltage. Behind each wall outlet is a complicated network of power lines which eventually leads to a central power station. Here steam turbines or water wheels turn huge generators which can supply thousands of amperes of current. Most of these power plants produce a.c. voltage, but there are still a few, in the older sections of force extending from the N pole to the S pole.

The coil of wire is mounted between the poles of the magnet in such a way that it can be rotated. Connections are made to the two ends of the coil by means of *slip rings* and *brushes*, as illustrated in Fig. 24. The copper slip rings are mounted on the shaft but insulated from it. The



FIG. 24. Simple a.c. generator. Yes, it works

of large cities, which generate d.c. voltages for home and business use.

A general idea of how these a.c. and d.c. voltages are produced is well worth while, because it will help you to understand how these voltages affect radio circuits.

A.C. Generator

In its simplest form, an alternating current generator has only the two essential parts pictured in Fig. 24, an electromagnet and a rotating coil.

Construction. The electromagnet in this elementary generator is a horseshoe-shaped iron piece which is magnetized by a coil of wire connected to a d.c. voltage source. The electromagnet produces magnetic lines brushes are blocks of carbon which are mounted in fixed holders. The brushes slide easily against the slip rings as the shaft rotates, and provide a complete path for electrons from the coil to any loads which are connected to the generator terminals.

Operation. When the coil is rotated, the number of magnetic lines of force which pass through the coil changes continually. This causes a voltage to be generated or induced in the coil.

Polarity. During one half of each complete revolution or turn of the coil, this voltage acts in one direction. During the other half of a revolution, the voltage reverses its *polarity* and acts in the other direction.



FIG. 25. One cycle of the voltage produced by a simple a.c. generator can be represented by the curve shown here. If the generator were being turned slowly enough so a zerocenter d.c. meter could follow its variations, the meter readings at nine different coil positions during a complete revolution of the generator coil would be as shown here.

If we connect between the brushes of our simple a.c. generator a special type of d.c. meter which has zero at the *center* of its scale, and take photographs of the meter at each 1/8 revolution while turning the coil slowly, the meter photographs might appear as shown across the top of Fig. 25. These photographs give a good idea of how the voltage is varying both in value and polarity, so let's study them a bit.

During the first half-revolution of the coil (from 0 to $\frac{1}{2}$), the meter pointer moves from 0 to the extreme right, and then back to 0. During the second half-revolution, the generated voltage has reversed its polarity, and the meter pointer moves from 0 to the extreme *left* and back to zero again.

Graph for an A.C. Generator. Now, if we changed these meter readings into the type of simple graph used by radio men, we would secure a curve like that in Fig. 25. Distances up or down from the horizontal line represent voltage values at different instants of time (at different parts of a complete revolution). The curve is above the horizontal line for one polarity, and goes below the line when the voltage reverses its polarity.

One Cycle. One complete revolution of our simple generator coil produces one cycle of a.c. voltage. For each additional revolution, the voltage changes would repeat themselves in the same way as shown for the first revolution in Fig. 25. Consequently, the curve represents one cycle of our a.c. voltage.

60-Cycle Power. If we turn our simple generator fast enough so it makes 60 complete revolutions in one second, there will be 60 of these voltage cycles per second, and we will have 60-cycle electric power like that used in most sections of this country.

When a generator is rotated fast enough to produce 60 voltage cycles each second, no ordinary meter can follow the instantaneous voltage varia-



FIG. 26. Simple d.c. generator. It'll work, tool

tions in each cycle. A cathode ray oscilloscope can follow fast changes, however, and for one cycle of a 60cycle voltage it would show on its screen exactly the same curved pattern shown in Fig. 25.

D.C. Generator

A direct current generator is nothing more than an a.c. generator with a *commutator* in place of slip rings, as



HAVING ONE COIL

FIG. 27. Example of a diagram which might be used by radio men to show how the voltage of a simple d.c. generator varies during one complete revolution of the coil.

shown in Fig. 26. A commutator is a device which automatically reverses the connections between the generator and its load at the exact instant during each revolution when the generated voltage begins reversing its polarity.

Polarity. The automatic reversal of connections by the commutator makes the terminals of a d.c. generator have *the same* polarity at all times. As a result, a diagram showing the voltage between the brushes of a d.c. generator during one complete revolution would appear as in Fig. 27.

Actual Construction. In an actual d.c. generator such as would be used in a power plant and in certain types of radio equipment, there are many coils mounted on the rotating shaft of the generator, instead of just one coil. When the voltage in one coil is dropping, the voltage in another coil is rising, and the voltages add together to give an almost constant d.c. voltage.

Filter. Even a small variation in d.c. voltage is objectionable in certain types of radio equipment, but this variation can be removed with a *filter* made up of coils and condensers.

Field Current. In commercial a.c. and d.c. generators, the electromagnet coil is known as he *field coil* because it provides the magnetic field.

In a d.c. generator, some of the direct current produced by the generator



Examples of generator units used with radio equipment. A—Dynamotor, a combination motor and generator operating from a 6-volt storage battery and generating the high d.c. voltages required by police radio equipment or public address amplifiers. B—Dynamotor which changes 32 or 115 volts d.c. to 115 volts a.c. C—Wind-driven generator used for charging storage batteries of farm radios. D—115-volt a.c. generator which can be driven by the fan belt of an automobile or truck. E—Combination 115-volt a.c. and 6-volt d.c. generator driven by a gasoline engine, used with high-power portable radio equipment.

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is used for the field coil, thereby eliminating the need for a battery.

A commercial a.c. generator always has a small d.c. generator mounted on its shaft or built right into the a.c. generator housing, to produce the di-



FIG. 28. Diagram of a power transformer similar to those used in a.c. radio receivers.

rect current required for the field coils of the a.c. generator.

Transformers

In a.c. circuits, the problems of securing a desired a.c. voltage value is greatly simplified because we can change an a.c. voltage to any desired higher or lower value by means of a *transformer*.

Here is a practical example: A radio receiver which is to operate from a 115-volt a.c. power line requires 5 volts for the filament of the rectifier tube, 6.3 volts for the filaments of the other tubes, and 700 volts for the power pack which is to furnish d.c. voltages to the various tube electrodes.

The power transformer for this receiver will have one primary winding which connects to the power line, and three secondary windings providing the three required voltages, as shown in Fig. 28. You thus see that a transformer can be made either to step up or to step down the line voltage.

Looking Ahead

The general knowledge of radio broadcasting and receiving which you gained from the first two lessons of your Course has now been reinforced by the highly practical information in this lesson.

You have learned a great deal about voltage, current and resistance in radio circuits. You have a working knowledge of voltage source, load and meter connections. You are familiar with the different kinds of voltage sources used in radio work.

You are ready, therefore, at this early point in your N.R.I. Course to take up a typical modern superheterodyne radio receiver. In the next lesson, you will find out what this receiver is like, how it is operated, what each part is, why and how the parts might fail, and what radio men would do to repair the receiver when various parts fail.

Not only will this next lesson prove clearly the importance of the material you have already studied, but it will carry you right ahead to still more interesting and more practical work. You will actually be learning some of the different ways of repairing receivers and radio parts.

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Lesson Questions

Be sure to number your Answer Sheet 3FR-2.

Place your Student Number on every Answer Sheet.

Send in your answers for this lesson immediately after you finish them, as instructed in the Study Schedule. Do this, and get the greatest possible benefit from our speedy personal grading service.

- 1. What type of voltage reverses its polarity regularly?
- 2. Name the *unit* of current which is equal to one thousandth of an ampere.
- 3. If you want to reduce the amount of direct current which is flowing in a radio circuit, what radio part should you insert in the circuit?
- 4. If you *increase the source voltage* in a complete radio circuit, what will happen to the current?
- 5. If you decrease the resistance in a complete circuit, what will happen to the current?
- 6. In what way would you connect together the filaments of radio tubes if all the filaments required the same current, and the available voltage was equal to the SUM of all the filament voltage ratings?
- 7. When can the filaments of radio tubes be connected in parallel?
- 8. Will the voltage drops across the resistors in this series circuit add up to a value *LESS THAN*, *EQUAL TO*, or *MORE THAN* the 45-volt source voltage?
- 9. What reading is indicated by the pointer on this voltmeter scale? (Study Fig. 13 carefully before you answer this question.)





10. Tell how you would connect four 1½-volt dry cells together to get a total voltage of 6 volts?



GETTING YOUR "SECOND WIND"

After an opening burst of speed, a champion longdistance runner drops down to a steady natural pace. This brief period of relaxation releases that reserve of power called "second wind." He is then able to overtake and pass the now nearly exhausted leaders in the race.

No matter what you are doing, you also can do it better once you get your *second wind*. Instead of fighting that sleepy feeling which sometimes comes when you are studying, stop and relax for a few minutes so as to release your own supply of reserve power. Here are some ways to relax, all worth trying.

Get up and exercise for a few minutes. Get a drink of cold water. Step outside for a few *deep* breaths of fresh air. Take a brisk walk up and down the road or once around the block.

If you "shake yourself awake" in one of these ways, you'll find it isn't at all hard to get that second wind which enables you to study longer and makes the going easier. Then you'll get in some real worthwhile studying—then you'll get things done!

J. E. SMITH.





STUDY SCHEDULE NO. 4

To master this fourth part of your N.R.I. Course most speedily and thoroughly, divide your study into the steps given below. First read the specified pages at your usual reading speed. Next, reread them slowly one or more times, endeavoring always to *understand* as well as remember. Finish with one quick reading to tie the ideas together in your mind, then write on scrap paper your answers to the questions specified for that step. After finishing the lesson, copy all answers neatly on an Answer Sheet.

- □ 2. Radio Receiver Tubes......Pages 6-9 Since a tube is one of the most important parts in modern receivers, you study first the five tubes used in the average superheterodyne receiver which serves as our example for this lesson. Tubes will mean a lot more after you finish this interesting section. Answer Lesson Questions 2 and 3.
- □ 3. Above-Chassis Parts.....Pages 10-16 You take up one by one the parts which are mounted on top of the chassis of our superheterodyne receiver—the loudspeaker, the output transformer, the gang tuning condenser, the trimmer condensers and the i. f. transformers. You find out how each one is constructed, what it does, how it might fail, and how it is repaired or replaced. Answer Lesson Questions 4, 5 and 6.
- □ 4. Under-Chassis Parts.....Pages 17-25 You turn the chassis upside-down, and study in turn the soldered joints, resistors, condensers, r. f. coils, power transformer, power cord, on-off switch and volume control, tone control, dial lamp and tube sockets. Answer Lesson Questions 7, 8 and 9.
- □ 5. Radio Receiver Servicing Techniques......Pages 26-28 Here's your first real introduction to actual servicing techniques. You learn how a radio man uses various techniques to find first the defective section or stage, then the defective circuit, and finally the defective part. Answer Lesson Question 10.
- □ 6. Mail Your Answers for Lesson 4FR-2 to N. R. I. for Grading.

□ 7. How To Remove and Overhaul the Chassis.......RSM Booklet No. 4 This is another of the series of Booklets that are to give you practical, how-todo-it training in fixing radios. Read this Booklet now, then keep it handy for future reference. The "shop training" you get from these RSM Booklets is going to prove valuable to you throughout your service career, so you will want to read and review these Booklets many times. NOTE: Some of the Study Schedules in future Lessons may refer to "Job Sheets" instead of RSM Booklets. Read the correspondingly numbered RSM Booklet in such cases. These Booklets are an improved and enlarged version of the older Job Sheets, and are replacing them.

8. Start Studying the Next Lesson, on Resistors.

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GETTING ACQUAINTED WITH RECEIVER SERVICING

We Study a Modern Radio Receiver

IF RADIO servicing is to be your chosen work, you will unquestionably deal with radio receivers most of the time. And no matter what other branch of radio you intend to get into, you will still work with receivers part of the time.

For example, in practically every broadcast station there is at least one radio receiver. This receiver is tuned to the station frequency, and enables the radio operator on duty to hear exactly what the thousands of radio listeners are hearing. A radio operator at a huge transmitter would certainly feel embarrassed if he had to call in a radio serviceman to fix his own monitor receiver when trouble developed.

The two-way radio communication stations used on boats, airplanes and other mobile equipment provide more examples. Failure of just one tiny part in the radio receiver can put one of these stations out of action just as effectively as failure of the largest part in the station's transmitter.

The fundamental principles which make a radio receiver work also apply to transmitters, test instruments, and all other radio apparatus. Yes, radio receiver parts and circuits work exactly like the corresponding parts and



Photo by F. C. Wilkinson

Courtesy General Electric Co.

Radio men use receiver servicing techniques for all types of radio equipment. Thus, the method which the man at the left is using to repair a modern all-wave superheterodyne receiver will apply equally as well to the television transmitter being serviced by the man at the right.

circuits you find in all other types of radio equipment, wherever you may be—in the radio room of an ocean liner, at the radio controls in a transport air liner, at the sending key of a forest fire lookout station, in the control booth of the public address system at a huge football stadium, or in any of the broadcast and short-wave radio stations throughout the world.

Once you learn how to repair radio receivers by using modern professional servicing techniques, you will have the foundation training for operating and repairing practically any other piece of radio apparatus.

Preview of This Lesson

In this lesson, you imagine that you have in front of you the table model receiver pictured in Fig. 1. It is an average modern receiver, using the latest superheterodyne circuit (pronounced SOO-pur-HET-ur-o-DINE, and often abbreviated as superhet or super). With the aid of special illustrations, you learn about the parts in this receiver and their general purposes, just as if the set were right in front of you in your own home.

Starting with the front of the receiver, you learn what each control knob does and how it is used. The meanings of the printed numbers and scales on the receiver tuning dial are clearly explained. You find out what to do when a customer loses one or more of the front-panel knobs or controls.

You remove the "works" (the chassis, pronounced CHASS-iss) from the receiver cabinet, then examine in turn each of the parts on top of the chassis, including the radio tubes, the loudspeaker, the output transformer, the gang tuning condenser, and some new parts which you take up for the first time here. In the same way, you study



FIG. 1. This modern superheterodyne receiver is used as an example throughout this lesson.

the many smaller parts which are located underneath the chassis.

Throughout this lesson, the practical viewpoint is stressed. You learn a great deal about how a part is made, what it does, how it can fail or give trouble, and how it is repaired or replaced.

Radio Map of a Superhet

Your study of this lesson will be much easier if you first get a general idea of what a superheterodyne circuit is. You are already familiar with the other common type of receiver circuit —the t.r.f. circuit, so it will be particularly interesting to find out why the superheterodyne can "run circles" around a t.r.f. receiver of the same size.

Figure 2 is the "radio map" which gives you the facts about the circuit arrangement of the superheterodyne receiver used as the example for this lesson. As you can see, the first stage in our receiver is the *MIXER-FIRST DETECTOR*. It receives two signals:

1. The desired modulated r.f. carrier signal, which is produced in the receiving antenna by radio waves. tion on your meter. Again, the a.c. resistance has no effect; the accuracy of the measurement depends only on the accuracy with which the capacity and the frequency are known.

COIL A.C. RESISTANCE

Almost always, the a.c. resistance in a tuned circuit will be in the coil. You can neglect the resistance in the average tuning condenser.

You can use the circuit shown in Fig. 10A to measure the coil resistance. First, set resistor R to zero, bring the circuit to resonance and note the current reading on the meter. Then adjust resistor R to a value which reduces the current meter reading to one-half its first reading. This value of R is equal to the a.c. resistance of the coil.

You can easily see why. At resonance, the coil inductive reactance and the capacitive reactance cancel each other, and the circuit current is determined only by the source voltage and the a.c. resistance. When resistor Ris increased to a value which cuts the current in half, the resistance in the circuit must have doubled, because you have not changed the source voltage. Therefore, this value of resistor R is equal to the a.c. resistance of the coil. Of course, this method requires that resistor R be variable and be calibrated at the frequency used in making the measurement.

If a variable resistor is not available, then a slight change in the method will allow you to use a fixed resistor to measure the coil resistance. To do so, first measure the current flow with the resistance out of the circuit. Call this I_1 . Then, insert a fixed resistor R in the circuit, remeasure the current and call the new reading I_2 . The a.c. resistance of the I_R

$$\operatorname{coil} = \frac{I_2 R}{I_1 - I_2}.$$

The fixed resistor should have a value reasonably close to the expected coil resistance, which may range from a fraction of an ohm to 10 or 15 ohms. If it is too large, I_2 will become a very small reading, difficult to accurately determine on the meter scale.

COIL Q FACTOR

As you know, the Q factor of the coil is equal to its reactance divided by its a.c. resistance. If you know the inductance, you can compute the reactance ($X_L = 6.28fL$), find the resistance by one of the methods just described and so find the Q of the coil at that particular frequency.

You can also find the Q of the coil with the circuit shown in Fig. 11. You



FIG. 11. A circuit for determining Q.

do not need to know the inductance or resistance of the coil, nor do you need a calibrated condenser. This circuit works on the basic principle of resonant voltage step-up, making use of the fact that, at resonance, the voltage across the condenser or the coil equals Q times the source voltage. Thus, you can divide the voltage across the condenser by the source voltage and obtain the Q of the circuit—which is the coil Q when the resistance is all in the coil.

You can easily measure the voltage across the condenser with an r.f. vacuum tube voltmeter, but measuring the source voltage in the resonant circuit is more difficult. The best way is to insert a known small resistance in the resonant circuit and use the voltage drop across it as the source voltage (Fig. 11). Resistor R has very low resistance (a fraction of an ohm) which is only a fraction of the expected coil resistance value. By measuring the current flow from the oscillator through R with the meter $I_{\rm M}$, you can find e, which equals $I_{\rm M} \times R$.

Tune the circuit to a maximum reading on the v.t.v.m., divide this voltage by e, and you have the Q of the circuit. While this does not give the a.c. resistance or coil reactance, it does give you a way to compare coils for merit. If



Courtesy General Raaso Interior construction of a standard variable condenser.

you know either the reactance or the r.f. resistance, you can find the other X_L from Q = -. Of course, the circuit

 $Q \equiv ---$. Of course, the circuit R

resistance found from this formula includes the resistance of R, but this is usually small enough to neglect. If you wish to be very accurate, you can subtract it from the total resistance found.

• Of course, the reactance and a.c. resistance both vary with frequency, so the Q found holds good only near the frequency used. If the Q at some other frequency (or over a band of frequencies) is required, then you will have to tune the oscillator to these other frequencies, retune for resonance, and divide the new value of E by the new e to get the Q at these points.

DISTRIBUTED CAPACITY

Occasionally you may want to know the distributed capacity between the turns of a coil. You can measure this with either of the circuits in Fig. 10. First, tune the circuit to resonance at some frequency which uses near maximum capacity of condenser C_8 . Call this capacity C_1 . Now, without changing the r.f. oscillator frequency setting, adjust C_8 to tune the circuit to the second harmonic of the oscillator. Record this value of C_8 (which will be about one-quarter the first value) as C_2 . The distributed ca- C_1-4C_2

pacity of the coil, C_0 , equals —

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Bridge Measurements

The bridge circuit is the most widely used means of measuring inductance and capacity values. It has the advantage of being a ratio device, so that the values of parts far different from the standard can be checked. Just a few standard parts permit a very wide range of measurement. Let's see how these circuits work.

THE WHEATSTONE BRIDGE

You have met the basic bridge circuit in other lessons. Essentially, it is a connection of four resistors, with a voltage supplied to two terminals and a meter between the other two terminals, as in Fig. 12. To see how the circuit works, let us presume that R_x is an unknown resistor, and that R_s is adjusted until a value is found which makes the meter read zero.

The fact that the meter reads zero means that points B and C are at the same potential. Tracing the circuit, you will see that this means the voltage drop across R_1 equals that across R_3 , and the drop across R_x equals that across R_2 . In other words, assuming that a current I_1 flows through resistors R_1 and R_2 , and that another current I_2 flows through R_3 and R_x ; the circuit is so adjusted that:

and

$$R_{\mathbf{x}} \times I_{\mathbf{z}} = R_{\mathbf{z}} \times I_{\mathbf{1}}$$
$$R_{\mathbf{x}} \times I_{\mathbf{z}} = R_{\mathbf{1}} \times I_{\mathbf{1}}$$

If you divide the upper equation by the lower equation, the currents will cancel and you will have:

$$\frac{R_{\mathbf{X}}}{R_{\mathbf{3}}} = \frac{R_{\mathbf{2}}}{R_{\mathbf{1}}}$$

This says that the ratio of R_x to R_s is the same as the ratio of R_2 to R_1 . Carrying this a step further, you have:

$$R_{\mathbf{X}} = \frac{R_{\mathbf{z}}}{R_{\mathbf{z}}} \times R_{\mathbf{z}}$$

This is a very important equation. It shows that if you make R_1 equal to R_2 , then R_x will be equal to the value of R_s which will bring the bridge into balance (give zero meter reading). But if R_1 and R_2 are not equal, this equation further shows that $R_{\mathbf{x}}$ will be equal to the ratio of R_2 to R_1 , multiplied by the value of R_3 which bal-This means that ances the bridge. even if resistor R_s has only a limited range of variation, you can still measure an unknown resistor $R_{\mathbf{x}}$ of almost any value merely by choosing the proper ratio between resistors R_1 and R_2 . This, of course, gives you a highly flexible measuring circuit.

In practical bridge circuits, resistors R_1 and R_2 are made so that they can



FIG. 12. The basic Wheatstone resistance bridge.

be varied in decimal ratios to each other—such as 1 to 1, 1 to 10, 1 to 100, 1 to 1000, or 1000 to 1, 100 to 1, or 10 to 1. Usually, R_1 is an accurate fixed resistor, and R_2 a series of fixed resistors arranged with a switch so the ratio can be varied in steps. Then the only continuously variable resistor will be R_3 , and even for this, fixed resistors in a decade box arrangement may be used.



FIG. 13. A decade box.

▶ As shown schematically in Fig. 13, a decade resistance box is a series of accurate resistances connected to contacts. Various resistances can be obtained by switching to various contacts. The name comes from the fact that each switch section has ten contacts, so there are ten steps (decade means a group of ten) in each section.

With the particular arrangement shown, any resistance between 0 ohms and 1110 ohms can be obtained in 1ohm steps. (The switches are shown set in the 0 ohm position.) If, for example, switch SW_1 is moved over to contact 4, there will be 4 ohms resistance between terminal A and terminal B. If SW_2 is then set to contact 6, 60 ohms will be added to the 4 ohms and you will have 64 ohms. If SW_3 is moved to contact 8, 800 ohms will be added in the circuit; you would then have a total of 864 ohms, etc. The values of the resistances need not be 1, 10 and 100 ohms each; they can be small fractions of an ohm or much larger values, depending on the design of the decade box.

▶ In using a bridge, the ratio of R_2 $\div R_1$ is adjusted to a likely value $(R_x \text{ may or may not be approxi$ $mately known) and <math>R_s$ is adjusted to make the meter read zero. If the bridge cannot be made to balance, other ratios are tried until the zero reading is found. Then the value of R_x is the $R_2 \div R_1$ ratio multiplied by R_s .

• Of course, the meter will read until the bridge is balanced. Since the amount of meter current depends on the amount of unbalance, it may be large if the bridge is far from balance. Further, the direction of current flow through the meter depends on whether the unknown is larger or smaller than the ratio times R_{a} .

In order to let the meter pointer swing in either direction, the meter springs are adjusted so zero is in the center of the scale. By noting which way the meter deflects, you can tell what adjustment is needed to balance the bridge. Thus, a deflection to the left may mean an increase in R_3 is needed for a balance, while a deflection to the right may mean R_3 should be decreased. It may be the other



FIG. 14. For meter protection, switch position 2 is used until the bridge is nearly balanced.

way around, depending on how the battery is connected to the bridge terminals.

To prevent ruining the meter with excess currents, a switch like SW in Fig. 14 is usually used. This is a threeposition switch that can be thrown to the right or the left from its normal resting position. Position 1 is the normal position and opens the meter circuit. (A spring returns the switch to this position when it is not held in one



FIG. 15. This basic a.c. bridge is just like the d.c. bridge except for a change in the voltage source and the indicator.

of the other positions by the operator.) In making readings, the switch is first thrown to position 2, closing the meter circuit and connecting a shunt across the meter at the same time. This makes the meter insensitive and protects it from excess current. The bridge is adjusted until the meter shows little or no deflection, then the switch is moved to position 3, which places the sensitive meter directly in the circuit, and a final balance is found.

Until the bridge is almost balanced, the switch is closed only long enough to see whether the meter pointer moves rapidly away from zero. Then, as the balance point is neared, the current drops so that a reading can be taken safely. Even so, the switch is allowed to return to position 1 while any adjustment is made.

A. C. BRIDGES

It is not necessary to use a d.c. power supply with a bridge. You can use an a.c., a.f. or r.f. voltage if you wish but you must use an indicating device suitable for the kind of voltage used. For an audio voltage, an a.c. meter or a pair of headphones may be used, connected in the position shown in Fig. 15. Balance will be shown by a zero meter reading or minimum sound in the phones. R.F. voltages require a v.t.v.m. as the indicator.

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Generally speaking, an a.c. bridge is more useful than a d.c. bridge, for we can measure coils and condensers as well as resistors with it. The a.c. bridge comes to balance when the *impedances* of the arms are in the proper ratios.

You might think that you have to use coils in all four arms of the bridge to obtain an inductance balance, and capacities in all arms for a capacity balance. The fact is, the ratio arms R_1 and R_2 can be resistances as long as they do not have appreciable inductance or capacity. Then a standard inductance can be used in place of R_3 for inductance measurements, or a_{γ}



Courtesy Industrial Instruments Co



standard capacity in place of R_s for capacity measurements.

Capacity Bridge. The bridge for capacity measurements shown in Fig. 16A looks just like the resistance bridge of Fig. 15 except for the substitution of the two condensers. This bridge will balance when the impedances of the condensers have the rela-

tionship $Z_{\rm CX} = \frac{R_2}{R_1} \times Z_{\rm CS}$.

We can find the capacity of C_x if we assume that the impedance of a condenser is approximately the same as its reactance (for air- or soliddielectric types the resistance is negligible) and remember that the reactance is inversely proportional to the capacity,

$$(X_{\rm C} = \frac{1}{6.28fC}),$$

Substituting this relationship in the above formula and simplifying, you

get $C_{\rm x} = \frac{R_1}{R_2} \times C_{\rm s}$. Notice that this

capacity formula has just the opposite ratio of resistances from the formula used with the resistance bridge.

Inductance Bridge. On the other hand, you should use the same resistance ratio for the inductance bridge (Fig. 16B) as for a resistance bridge, because inductive reactance is proportional to inductance. The formula for

this bridge is:
$$L_{\rm X} = \frac{R_2}{R_1} \times L_8$$

Fixed Standards. In the circuits shown in Fig. 16, the standard parts are variable. Each circuit is used just like a resistance bridge, with ratio arm adjustments made if the range of the variable is not great enough to balance the bridge. Often, however, variable standards are not available. If they are not, fixed coils or condensers can

be used as L_8 or C_8 and the ratio arms can be adjusted to balance the bridge. When this method is used, R_2 should be a continuously variable resistor rather than a step or decade box type.

While theoretically you need only one fixed standard part for this bridge, in practice you will find a fine balance cannot be easily obtained, and the bridge becomes inaccurate if the ratio between R_1 and R_2 becomes very large. Therefore, it is usually best to have several fixed standard parts available and use the one which gives a balance most nearly in the ratio of 1 to 1. If the value of the unknown part cannot be guessed, use a fixed part of any



FIG. 16. The bridge at A is a capacity bridge, while B shows an inductance bridge.

value and determine the approximate ratio. Then use a smaller or larger part to bring the ratio down to a point where a fine balance may be easily obtained.

The fact that a fixed inductance or

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capacity can be used in the bridge makes the instrument very practical. Fixed standards are comparatively cheap and readily available. For ordinary service work, readily obtainable parts of 2% to 5% tolerance may be entirely adequate for checking the radio parts-which often have tolerances as high as 20%.

Wagner Ground. When you measure high-impedance parts (small-



FIG. 17. Balancing to ground to eliminate body capacity effects.

capacity condensers or high-inductance coils), particularly at high frequencies, you may find that body capacity unbalances the bridge. For example, putting on the phones used as an indicator—or even bringing your hand close to the bridge—inay change your readings. Grounding the bridge directly would solve this difficulty, but would also present us with another one just as bad, for the stray capacities C_1 and C_2 shown in Fig. 17 would then shunt the ratio arms and would themselves determine the ratio setting.

The best solution is to balance the entire bridge with respect to ground. One way to do so is to use a system known as a Wagner ground, shown in Fig. 17. A low-resistance potentiometer R_s is connected across the bridge and the variable arm is grounded. As you see, this puts a low-resistance shunt across C_1 and C_2 . To operate a bridge using this ground, first balance the bridge in the usual manner as well as possible. Then, throw switch SW to the ground position and adjust R_{s} to give a minimum indication. (Notice that grounding one terminal of the indicator through switch SW makes another bridge of R_1 , R_2 and the two sections of R_s , as this connects the indicator between C and ground, and through the ground to the arm of R_{s} .)

The bridge is balanced with respect to ground when a minimum indication is obtained. You should then throw the switch back to its normal position and readjust the bridge to give a minimum indication. If necessary, repeat this process until further adjustments cause no upset of the bridge.

Since the trouble is caused by capacity to ground, the resistance of R_{\bullet}



FIG. 18. Using R permits the bridge to be balanced for the resistance of the unknown part while its reactance is balanced by that of the standard part. This gives a direct measure of the a.c. resistance of the unknown part and permits the calculation of the Q.

may be too high to permit a fine balance at the higher radio frequencies. For high-frequency work, some bridges use the split stator condenser C_a shown



Using a capacity bridge of the type having a tuning eye indicator. This bridge will measure the power factor as well as the capacity.

in Fig. 17. This is a condenser with two stator sections and a single common rotor section; as the condenser is rotated, the capacity of one section increases while that of the other decreases. Its capacity can thus be varied to make the capacitive reactance between A and ground and that between D and ground have the same ratio as the $R_1 \rightarrow R_2$ ratio, thus balancing the bridge with respect to ground. The balancing procedure is exactly like that used with the Wagner ground—except, of course, you vary C_8 instead of R_8 .

IMPROVED BRIDGE CIRCUITS

The basic bridge we have discussed so far is excellent if you can neglect resistance in the coil or in the condenser being checked and the resistance in your standard. But a zero current indication can be obtained only if the *impedances* of the arms are exactly equal; if the unknown has appreciable resistance as well as reactance, you can get a *minimum indication* but not a true *zero indication*. Also, this causes some error in adjusting the bridge. However, a rather simple means can be used to get a true null (zero) balance and, at the same time, determine the Q factor of the coil or condenser.

Usually the standard coil or condenser is chosen to have very low resistance, so if a resistance unbalance exists, it will usually be the fault of the unknown coil or condenser. As shown in Fig. 18, you need only add resistance in series with the standard coil or condenser to bring the bridge back into balance. To use this circuit, insert the coil or condenser standard and the unknown as indicated, and adjust the standard or the ratio arms to give a minimum indication. Then, adjust Rfor a further reduction. Repeat the
standard and R adjustments until the indication cannot be reduced further. When the *least possible* signal is heard or indicated, the reactive components of the standard and unknown are balanced, and so are the resistive components of the unknown and resistor R. Multiplying the value of resistor R by $X_c = \frac{1}{6.28fC}, \text{ so } Q_c = \frac{X_c}{R} = \frac{1}{6.28fCR}.$ As the power factor is approximately $\frac{1}{Q}$ for low power factor condensers,

the power factor is 6.28 fCR.



FIG. 19. A simplified diagram of a typical capacity bridge of the type made for radio servicemen. The bridge "corners" are A, B, C and D. Condensers C₁, C₂ and C₃ are the standard parts, selected by SW₁. Resistor R₁ makes up the A-C and C-D legs of the bridge. When it is varied, it adds resistance in one leg and takes it out of the other at the same time. Condensers C₄ and C₅ are just blocking condensers to isolate the indicator, which is a magic eye tube VT operating from the voltage across R₂. An adjustable d.c. supply is at the right, for leakage tests. To use, the unknown condenser is connected to C_x and SW₂ is thrown to the right to apply the rated d.c. from SW₃. The neon bulb will flash or glow, the rate of flash depending on the leakage. Assuming a satisfactory leakage value (and no short-circuit causing a continuous glow), SW₂ is thrown to the left, SW₁ is set to use the proper standard, and R₁ is adjusted for a maximum closure of the eye indicator, indicating a minimum voltage across R₂. Should the adjustment of R₁ approach but not pass through a balance when moved all the way to the A end, the wrong standard is selected by SW₁ or the C_x condenser is open. The same indication at the D end shows the wrong standard or a shorted C_x. When checking large condensers, the SW₁ setting including R in the circuit is used, and R is adjusted for further closure of the eye. This resistor is calibrated to show the power factor directly.

the bridge ratio will give the amount of resistance in the unknown part.

Thus, you can determine the capacity or inductance and the a.c. resistance of the unknown part at the same time. If you want to find the Q factor of an inductance, find the inductive reactance from the formula $X_L =$ 6.28 fL, then divide this reactance by the resistance. Similarly, you can find the Q of a condenser by dividing the reactance by the a.c. resistance. However, the reactance of the condenser is ▶ The circuit shown in Fig. 18 is the basis for several capacity bridges designed for radio servicemen. A typical instrument is shown in Fig. 19. By-calibrating R in terms of power factor and inserting a d.c. power supply to polarize electrolytic condensers, a bridge of this kind can be used to measure capacities, leakage and power factor for condensers ranging from the smallest air and mica types up through large electrolytic condensers. $\sqrt{}$ The capacity and power factor are in-

dicated directly on a calibrated scale.

Comparing L with C. In the basic bridges covered so far, coils have been compared with coils and condensers with condensers. There are many variations of these bridges, found mostly in laboratories. Some have resonant circuits in their arms while others are designed for certain special purposes.

Another type, which balances a condenser against a coil, is very useful. A basic form of this bridge, using a standard condenser (which is more likely to be available than a standard coil) is shown in Fig. 20. Since the



FIG. 20. This bridge balances a capacity against an inductance.

capacitive reactance is opposite in phase to the coil reactance, the balancing condenser must be in the position shown. The value of

$L_{\mathbf{X}} = R_2 R_3 C_1.$

Fig. 21 shows the Owen bridge, a more common form of this circuit, which also takes the a.c. resistance of the coil into account. The capacity C_2 is used to balance the coil resistance

 $R_{\mathbf{x}}$. The a.c. resistance of the coil is found from the formula

$$R_{\mathbf{x}} = \frac{C_1}{C_2} \times R_s,$$

while the inductance is found by the formula in the preceding paragraph.

This bridge is easily converted into a standard capacity bridge, so it can be used to check both inductance and capacity, using only a standard condenser.

Bridge Limitations. Most a.c. bridges work over a limited audio to low r.f. range. If the frequency is



FIG. 21. The Owen bridge.

increased, complete shielding of each element becomes necessary, and many precautions must be observed, to prevent stray fields and coupling between parts from upsetting results. Although high-frequency r.f. bridges are available, they are costly and are found only in large laboratories. For ordinary purposes, the resonance method is usually used when very high frequencies are involved. 2. The r.f. signal, which is selfgenerated in the R.F. OSCILLATOR stage in such a way that its frequency is exactly 456 kc. higher than the carrier frequency of the desired incoming signal.

Now comes the action which makes the superhet so different from a t.r.f. receiver. The mixer-first detector combines its two incoming signals to give an entirely new signal called the *i.f. signal.*

The new modulated i.f. carrier sig-

After the carrier frequency of the desired incoming signal has been changed to 456 kc. by the mixer-first detector, the resulting i.f. signal is fed into the *I.F. AMPLIFIER*. There it receives a tremendous amount of amplification, because an i.f. amplifier stage which works at only one carrier frequency is many times more effective than an amplifier stage which must work at many different frequencies. (In a t.r.f. receiver, you know, the r.f. amplifier stages must handle any car-



FIG. 2. Arrangement of stages in the superheterodyne receiver used as an example in this lesson.

nal in this receiver (and in many other superheterodyne receivers) *always* has a carrier frequency of exactly 456 kc. (This value is the difference between the frequencies of the r.f. signal and the incoming carrier signal.)

Strictly speaking, we should call this new signal the *modulated intermediate frequency carrier signal*, because it still has the original audio signal modulation. Since you are going to work with radio men, however, you will undoubtedly want to talk their own language, and say "eye ef signal" just as they do.

The frequency of the i.f. carrier signal is *in between* the r.f. carrier frequency and audio frequencies. This is why the new signal is called an *intermediate* frequency signal. rier frequency which the listener tunes in.)

The strengthened i.f. signal enters the SECOND DETECTOR. This stage separates the audio signal from the i.f. carrier signal, and also gets rid of the i.f. carrier since it is no longer needed. The second detector thus gives us the audio signal we have been working for all this time.

The audio amplifier and loudspeaker, following the second detector of this super, are exactly like the audio amplifier and loudspeaker of a t.r.f. receiver. Thus, the audio signal gets one boost in strength from the *FIRST A.F. STAGE*, and gets another boost from the *AUDIO OUTPUT STAGE* before being fed to the *LOUD*-*SPEAKER*. Here the audio signal

Frequency Measurements

The number of times an event repeats itself within a specified length of time is its frequency—usually measured in cycles per second. Therefore, the first requirement for accurate measurement of frequency is a highly accurate way to tell time. The fundamental unit of time is determined by the rotation of the earth. By observing the positions of certain stars. astronomers determine this period From these observavery exactly. tions extremely accurate standard Now if you clocks are calibrated. were to operate an oscillator in such a way that an electric clock is driven from its output, you could compare the time kept by the electric clock with that kept by a standard clock and have a very precise way to measure both the frequency of the oscillator and its ability to maintain that frequency. You could then use the oscillator output as a standard of frequency.

Primary standards of frequency of this kind are found in the National Bureau of Standards and other large laboratories. They are crystal oscillators, because this type has the great-



FIG. 22. How an electric clock is run from a crystal oscillator.

est frequency stability, and are kept at constant temperatures to make them as stable as possible. \bigvee

Usually these oscillators produce frequencies of 50 or 100 kilocycles. Either frequency is too high to drive an ordinary clock, so it is stepped down by a chain of multivibrators. (The multivibrator is a circuit which

can produce an accurate fraction of a frequency fed into it.) Fig. 22 shows how the crystal oscillator is used to drive a chain of multivibrators which finally produces a signal capable of driving a clock. The clock used follows frequency exactly, so comparing its time with that of a standard clock shows us the frequency of the crystal oscillator and also indicates any variations in this frequency. By making adjustments to keep the clocks together, the output of the oscillator can be used as a frequency standard.

The Multivibrator. This interesting circuit does not "vibrate"---it is



FIG. 23. A multivibrator is a resistance-coupled amplifier which can oscillate.

an oscillator consisting of a two-stage resistance-coupled amplifier like that shown in Fig. 23. Notice that condenser C_2 provides a feedback path from the plate circuit of the tube VT_2 to the grid circuit of tube VT_1 . Each tube reverses phase by 180° so two tubes produce a 360° reversal. This means the voltage across R_4 is back in phase with the input of VT_1 , so this feedback path allows the circuit to oscillate.

▶ By inserting a bias, we can make one tube draw slightly more plate current than the other when the circuit is first turned on. For this explanation,

we will assume that VT_1 draws more current than VT_2 . The initial rush of current through R_2 produces a voltage pulse across it, with a polarity such that the end of R_2 marked 1 is negative. This pulse passes through C_1 and is developed across R_3 , driving the grid of VT_2 negative and so cutting off its plate current. This causes a rapid decrease in the voltage drop across R_4 ; you can consider this decrease to be a voltage pulse with a polarity such that the end of R_4 marked 3 is less negative (more positive than normal). This pulse, in turn, passes through the coupling condenser C_2 and is developed across R_1 , making the grid of VT_1 positive and so increasing its plate current.

The plate current of VT_1 increases until the tube becomes saturated When this occurs, the current becomes constant, and so does the voltage drop across R_2 . In other words, there is no longer a voltage pulse across R_2 and, naturally, no pulse voltage developed across R_s . With the disappearance of the pulse voltage across R_3 , the grid of VT_2 is able to go back to its initial bias. (This does not happen instantaneously, however, because C_1 has been charged and must discharge through R_s before the grid of VT_2 can reach its original bias; this discharge, of course, takes a little time.)

As the grid of VT_2 approaches its initial bias, plate current again starts to flow through the tube. This increasing current creates another voltage pulse across R_4 . Because the current is increasing, this time the pulse has a polarity such that the end of the resistor marked S is negative. This pulse, passing through C_2 , drives the grid of VT_1 negative, and stops the flow of plate current through the tube. This causes a voltage pulse across R_2 , with a polarity such that point 1 is more positive; passing through C_1 , this pulse makes the grid of VT_2 positive, and the current through VT_2 increases.

Now it is the turn of the current through VT_2 to increase until saturation is reached. When the tube does saturate, VT_1 is able to pick up again, while VT_2 shuts off. This process is repeated over and over, with current flowing through first one tube, then the other. The net effect of this action, as far as output is concerned, is that voltage pulses appear across R_4 .

▶ The rate at which these pulses occur—that is, the frequency of oscillation of the circuit—depends on the time constants of the C_1 - R_s and C_2 - R_1 combinations. As you just learned, C_1 must discharge through R_s before VT_2 can pick up; similarly, C_2 must discharge through R_1 before VT_1 can pick up. By adjusting the capacities and resistances of these combinations, you can obtain almost any fundamental frequency you want—from a very low one to a frequency well up in the r.f. range.

The output of this oscillator is rich in harmonics—in fact, the output wave may be triangular or even square in shape because there are so many.

Although the multivibrator 18 fairly stable, its frequency can shift somewhat with changes in tube characteristics, operating voltages, etc. However, it will "lock in" with control pulses fed in the input, and will then produce its fundamental just as accurately as the control pulses are maintained. Equally as important it will lock with a control pulse which is some multiple of its fundamental. For example, a 1-kc. multivibrator can be controlled by a 10-kc. control unit, as it will ignore all pulses in the input except the one it is locked with.

You see, at some time during the cycle of operation the voltage on the



Courtesy General Radio

This rack contains a frequency standard, employing a 50-kc. crystal and two multivibrators. By checking the clock against standard time signals, the accuracy of the frequency standard can be determined. By using the multivibrator outputs, frequencies from one cycle per second up to several megacycles can be obtained.

grid of VT_1 has such a value that the tube can almost conduct current. Now suppose you feed in a 10-kc. signal across R_1 . This signal adds to the voltage already present across R_1 . If you adjust the amplitude of this signal carefully, you can make it increase the voltage on the grid of VT_1 just enough to make the tube start conducting a tiny fraction of a second ahead of the time it would have started

if the signal had not been fed in. The circuit then proceeds through its normal operating cycle, unaffected by the input voltage, until it reaches the precise point where the input voltage is again - able to -make - VT_{τ} -conduct-Since the fundamental frequency of the circuit is 1 kc., this point will not arrive until 9 cycles of the input signal have passed by, as these intermediate pulses, when added to the voltage across R_1 , are not of sufficient amplitude to start tube conduction. However, the feedback voltage is continuously building up, and the tenth cycle of the input signal will make VT_1 conduct again. Thus, the multivibrator locks in step with every tenth cycle of the input frequency, but ignores the cycles in between.

▶ In other words, the oscillation of the vibrator is precisely controlled by the input voltage, which overcomes the tendency of the circuit to wander and shift frequency. Thus, the fundamental and all its harmonic frequencies will be just as accurately controlled as is the input voltage. Further, the circuit can be controlled by frequencies as high as the tenth harmonic of the fundamental. Therefore, you can lock the circuit at the input side with one of its harmonics, but use the fundamental as the output frequency, and so get an output which is any fraction (up to, say, a tenth) of the input frequency.

It is this characteristic that makes the multivibrator useful in the circuit in Fig. 22. The crystal produces a 100-kc. voltage which controls the 10-kc. (fundamental frequency) multivibrator. The 10-kc. multivibrator in turn controls a 1-kc. multivibrator. By bringing out taps from the multivibrators in Fig. 22, you can obtain voltages with frequencies maintained just as accurately as the crystal oscillator frequency itself. ▶ Because it has so many harmonics, you can obtain from the 10-kc. multivibrator frequencies of 10 kc., 20 kc., 30 kc., etc., all the way up to 1000 kilocycles. Similarly, the 1-kilocycle multivibrator produces 1-kilocycle steps



Courtesy General Radio

An engineer reading the clock on a primary standard of frequency.

all the way up to 100 kilocycles. If you wish, you can feed these outputs into other multivibrators and so get a large number of extremely accurate frequencies.

Then, the output from such a primary frequency standard can be used to calibrate secondary standards such practical radio devices as signal generators, wave meters and other frequency determining devices. Let us now see just how such secondary standards are used in practical radio work.

THE WAVE METER

A very common frequency-indicating device is a wave meter. This is nothing more than a coil-condenser resonant circuit. Many have some means of indicating when they are tuned to resonance—such as a vacuum tube voltmeter, a thermocouple current meter in series with the resonant circuit, or a pilot lamp (which will burn brightest when resonance is reached).

When the wave meter has no indicator, you must depend on the meters associated with the source as indicators. Practically all r.f. power oscillators and power amplifiers have plate current meters. When a wave meter is brought near a resonant circuit and is itself tuned to resonance, it will absorb energy from the source. The plate current rises when this power is being furnished, so the point of maximum rise in plate current is the point of resonance; provided you are not too closely coupled, so as to cause a broad response or stop the oscillator.

▶ To use a wave meter, bring its coil close to an inductance in the circuit producing the frequency to be checked. If it is physically impossible to do this, feed the output of the circuit into a coil and bring the wave meter near this coil. Then, vary the tuning condenser dial until resonance is indicated and read the frequency off



Courtesy General Radio

The condenser for this wave-meter is inside the box, while the round black object at the side is one of the plug-in coils. A pilot lamp on the panel is used as the resonance indicator. The condenser dial is directly calibrated in frequency, different scales applying to the different coils. the dial. You may have to use a calibration chart if the condenser dial is not calibrated directly in frequencies.

Wave meters will maintain their accuracy of calibration for quite a long time if made from precision parts.



Courtesy General Radio

A frequency monitor used to check the frequency deviation (or frequency change) of a transmitter. An extremely accurate crystal oscillator produces a practically constant frequency. The crystal output is mixed with the transmitter carrier and the resulting beat frequency operates the deviation meter. A transmitter shift of one-half cycle can be observed on the meter, allowing the operator to keep the transmitter accurately adjusted to its assigned frequency. The use of this kind of equipment makes the carrier frequency of broadcast transmitters useful sources of known frequency for calibrating radios and service type signal generators.

However, a wave meter must be carefully made to be at all accurate—and even then will not show small variations in frequency less than 1% of the fundamental. For this reason, they are now used primarily for experimental purposes where an approximate frequency measurement is all that is required.

FREQUENCY COMPARISON

The radio serviceman often makes frequency comparisons. The signal generator normally available for service work is accurate enough for all practical alignment purposes, but not for precision measurements if they should be necessary. Furthermore, the service instrument may drift, so its frequency scales are usually not accurately calibrated after a period of time.

However, a signal generator may be easily recalibrated by comparing its output with the frequencies of broadcast stations. These stations make handy frequency standards, since they are required by law to maintain their frequencies within 20 cycles of the assigned frequency.

All you need to recalibrate a signal generator (or any other oscillator with a frequency in the broadcast range) is a receiver and a knowledge of the exact frequencies of several stations. which you can find in any log book. Tune the set to a particular broadcast station as accurately as possible, using an output indicator. Then, tune the signal generator near the same frequency and feed its output into the receiver at the same time as the incoming program. You will hear a high-pitched squeal which will change to a lower and lower frequency as you tune closer to the broadcast frequency. If the signal generator is tuned past the broadcast frequency, the pitch will rise from a low-frequency to a high-frequency squeal.

At the point where the squeal is of the lowest frequency or disappears entirely, the frequency of the signal generator is identical with that of the broadcast station. This is known as the zero beat method of determining frequency.

After finding the zero beat point, all you need do is mark the signal



Courtesy General Radio

Using a heterodyne frequency meter to check a transmitter frequency. This device uses a stable variable oscillator and a crystal oscillator. The crystal and its harmonics are beat against the variable oscillator. The zero beat points, as indicated in the headphones, are the points where the frequency of the variable oscillator is exactly the same as the crystal or a crystal harmonic. Thus, the crystal oscillator serves to check the calibration of the variable oscillator and allow corrections to be made. Then, the output of the variable oscillator can be beat against the output of any other oscillator or transmitter within its range and will indicate the frequency of this transmitter or oscillator.

generator dial or adjust its trimmers so the frequency marking corresponds with the frequency of the broadcast station to which you are tuned. Then repeat the process with other broadcast stations, and you will be able to calibrate the signal generator (or oscillator) dial very accurately for broadcast frequencies.

Using Harmonics. By using harmonics of the signal generator output, you can calibrate it over the frequency bands which are lower in frequency than the broadcast band, and thus have available an accurately calibrated signal generator for making intermediate frequency alignments.

You recall that if an oscillator produces a fundamental frequency of 300 kilocycles, its harmonics will be at 600 kilocycles, 900 kilocycles, 1200 kilocycles and 1500 kilocycles. The 300-kc. frequency is below the broadcast band, but all of the above-mentioned harmonics are within the broadcast band. Therefore, if you tune the receiver to a broadcast station which has one of these particular frequencies, then tune the generator to exactly 300 kilocycles, you will get a zero beat from the harmonic of the oscillator beating with the incoming signal. (If the receiver dial is calibrated accurately, we need only tune in the signal generator harmonics at the proper points on the dial. However, the zero beat method is more accurate.)

In a similar manner, other points in the frequency range from 100 to 500 kilocycles can be checked by beating harmonics against known broadcast station frequencies.

▶ You should tune in harmonics at more than one place on the dial, unless you are certain of the fundamental frequency within a few kilocycles. For example, if you set the receiver dial at 900, harmonics of 90, 100, 112.5, 128.5, 150, 180, 225, 300 and 450 kc. would all cause zero beats. You must tune away from 900 and see where the next signal comes in, to identify the fundamental frequency accurately. Thus, if you get signals at 900 and 990 kc., the fundamental must be 90, because this is the only one of this group which would have harmonics coming in at these two points. Similarly, harmonics of 100 would come in at 900, 1000, 1100 kc., etc.

Remember, you can find the fundamental frequency by locating any two consecutive points on the dial where harmonics can be heard, and then finding the frequency difference between these dial settings. This difference is equal to the fundamental frequency of the signal generator.

You will have to be careful to identify each and every harmonic to be sure that you have the right fundamental. Thus, if you use 900 kc. as one setting and find another signal at 1350, you'll have to be very certain there is no signal in between. Harmonics of 112.5, 150, 225 and 450 kc. will all come in at both 900 and 1350 kc.

If you hear no signal between 900 and 1350, the fundamental must be 450 kc., because 1350 minus 900 is 450. In other words, you have picked up the second and third harmonics of this frequency. On the other hand, if there are intermediate signals between 900 and 1350-kc. settings. you must determine where they are and must find the frequency difference between any two consecutive ones to find the fundamental frequency.

► This method lets you correct or calibrate a signal generator with sufficient accuracy for most measurement purposes. If the procedure is followed carefully, the accuracy will be that of the broadcast station fre-

quencies, which are usually accurate within one part in 100,000, or better, so the inaccuracy will be within a small part of 1%.

Broadcast stations themselves use accurate means of frequency determination to maintain their frequencies within the above-mentioned limits. Naturally, this maintenance of frequency is very important—for, if they drifted about, they might interfere with each other.

DETERMINING FREQUENCY OF AN UNCALIBRATED OSCILLATOR

The method just discussed can be used for other purposes than correcting an approximately calibrated oscillator. You can also use it, for ex-



FIG. 24. A vibrating reed is shown here. A somewhat similar arrangement can be used to operate a tuning fork, which is a two-pronged "fork" made so its mass and springiness cause it to be mechanically resonant to some desired frequency.

ample, to calibrate an oscillator which is not calibrated at all.

To do so, just feed the output of the oscillator into a receiver and set the oscillator dial to a zero beat setting with a broadcast station, or depend on the receiver dial indications if it is accurately calibrated. Then tune the receiver slowly to higher frequencies and find the next zero beat setting. The difference between the two receiver frequencies which give successive zero beat indications is, of course, the fundamental frequency of the particular oscillator setting you chose.

By marking this oscillator dial setting with the frequency found, then repeating the process for other oscillator settings, you will be able to calibrate your oscillator accurately. Of course, as you learned in the previous section, you must be careful not to miss any harmonics as you tune the receiver.

AUDIO FREQUENCY MEASUREMENTS

The standard frequency source shown in Fig. 22 can be used to check audio frequencies by adding another multivibrator or two to the chain. This will give frequencies of 100 cycles or 1 cycle per second, depending upon whether one multivibrator or two are added, which can be used to calibrate an audio signal generator or an audio wave meter. Either of these devices can then be used to determine the frequency of an audio voltage.

► There is another source of standard low frequencies—a vibrating reed or tuning fork. These devices have a natural frequency of vibration which depends on mass and springiness. When forced into vibration by an electromagnetic device like that shown in Fig. 24 the fork or reed will vibrate with the greatest amplitude when the frequency of the voltage applied to the coil is the same as the natural frequency of the device.

This fact has been used as the basic principle of a simple frequency indicator, using the less-expensive reeds. Several reeds, each with a different natural frequency, are mounted so that they are driven by an electromagnet. When an a.c. voltage is fed into the electromagnet, the reed with the natural frequency closest to the frequency of the a.c. vibrates with the greatest amplitude.

Of course, such a device would re-

quire a great many reeds to cover the entire audio range, because a different reed would have to be used for every few cycles. It is therefore usually used only to measure frequencies within a small range, particularly those near power line values. As you know, the standard a.c. power line in the United States is a 60-cycle line. Power companies use vibrating-reed meters a great deal in checking the line frequency so as to keep it accurately controlled. (Incidentally, the accuracy of electric clocks used in





Courtesy Triplett Elec. Inst. Co. FIG. 25. An interior view and a face view of the vibrating reed type frequency indicator.

World Radio History

homes depends entirely on the accuracy of the power line frequency. The fact that they keep such good time shows how well this frequency is controlled.)

A typical vibrating-reed meter is shown in Fig. 25. The ends of the reeds are painted white, so you can see when they vibrate. Fig. 26 shows



FIG. 26. How the reeds indicate the frequency. As the 60-cycle reed appears largest, the frequency is 60 cycles.

its appearance when a frequency of 60 cycles is being measured. The 60cycle reed appears the largest because it is vibrating at its natural frequency and so has a large amplitude of vibration. The reeds on each side are also vibrating, but not as much.

▶ While we are on the subject, another type of power line frequency meter is shown in Fig. 27. This meter has two coils $(L_1 \text{ and } L_2)$, with a magnetic vane mounted so it is free to turn in their fields. When an a.c. voltage is applied to the terminals, current will flow through both coils. The external circuit parts R_1 , L_3 , R_2 and L_4 are so proportioned that equal currents flow in L_1 and L_2 when the applied a.c. has a 60-cycle frequency. The normal tendency of the magnetic vane in the meter is to line up along the axis of one or the other of the coils, but since both have equal fields, it stands still, halfway between the two axes.

Changes in applied voltage will not move the vane, because the voltage drops will merely increase or decrease in proportion in both halves of the circuit. However, as you know, coil reactance goes up with frequency. So, if the frequency increases, the drop across L_3 will increase, producing a greater current through coil L_1 ; at the same time, the reactance of L_4 will increase, allowing less current to flow through L_2 . Since L_1 now has the stronger field, the magnetic vane will swing to the right to line up more nearly with the axis of coil L_1 . On the other hand, if the frequency decreases, opposite effects occur and the vane will move to the left.

Properly designed, this instrument will measure very small frequency changes. A picture of a typical instrument is shown in Fig. 28.

▶ While vibrating-reed and magnetic



Courtesy James G. Biddle Co.

A portable laboratory type vibrating reed frequency indicator. Note the large number of reeds, providing a wide frequency range in one-quarter and one-half cycle steps. vane meters are found mostly in power stations, many laboratories also use them to check audio frequencies within their ranges.

THE WIEN BRIDGE AS A FREQUENCY INDICATOR

Any resonant bridge circuit can, of course, be used like a wave meter to determine frequencies. For audio frequencies, a bridge known as the



FIG. 27. The magnetic vane frequency meter.

Wien bridge (which is also very popular for measuring capacities) is particularly useful. Its circuit is shown in Fig. 29.

This bridge can measure frequencies because of the actions of the arms containing the condensers. Notice C_1 is in shunt with R_3 , while C_2 is in series



Courtesy Weston

FIG. 28. A face view of a typical magnetic vane frequency meter.

with R_4 . As you can see, the impedance between terminals A and C can drop to a very low value at high frequencies because of the drop in reactance of C_1 . However, no matter how high the frequency, the impedance between C and D can never drop below the value of R_4 . On the other hand, if the frequency is decreased, the impedance between A and C will never go higher than the value of R_3 , while that between C and D can go on up to practically an open circuit.



FIG. 29. The Wien bridge used as a frequency indicator.

Therefore, as you move from some one particular frequency in either direction, the reactances of the two arms will behave in different manners. In fact, they act much as a coil and condenser would—therefore, in effect, this is a resonant circuit. We won't go into the engineering behind this bridge, but if resistor R_1 is twice R_2 , C_1 equals C_2 and R_3 equals R_4 , the frequency of balance of the bridge will be 159.000

f = ---, where f is in cycles, R C

C is in microfarads and R is in ohms. In this case, R is the value of either R_3 or R_4 , and C is the value of either of the condensers.

is changed into SOUND WAVES which are the desired radio program.

You See How Radio Men Work. After becoming thoroughly acquainted with the various parts in this receiver, you will see how radio servicemen go about locating and repairing trouble in the receiver.

Once the trouble is located, the defective part is easily replaced or repaired. It is in finding the cause of the trouble that a complete knowledge of the purposes and actions of radio parts counts.

Front-Panel Parts

Control Knobs. The general arrangement of the controls on this receiver is typical of that employed in most modern receivers, and is shown in Fig. 3.

To turn on this set you turn the volume control knob clockwise until a click is heard. This turns on the power pack, by completing the circuit for the power cord which goes to a wall outlet.

Further clockwise rotation of this knob increases the volume of loudness of the program coming from the loudspeaker.

When you tune in a station with the *tuning control* knob, you are doing two things: 1. Rotating the gang tuning condenser to the correct setting for reception of the station; 2. Moving a pointer to the position on the tuning dial at which the frequency of the station is indicated.

The tone control knob allows you to change the tone of the sound coming from this receiver, by reducing the strength of higher audio frequencies. The lower audio frequencies will then predominate, and give a mellow bass tone which emphasizes drums, large bass horns and men's voices.

On some receivers, it is also possible to set the tone control to reduce the strength only of the *lower* audio frequencies. The high frequencies of violins, cornets, bells and women's voices will then be emphasized. Tone controls are provided because people's tastes differ regarding tone just as much as tastes differ regarding food, clothes, or almost anything else.

Tuning Dial. It is common practice among receiver manufacturers to mark the tuning dial in kilocycles, with the last zero in the frequency value omitted. The dial shown in Fig. 3 is an example of this practice. If you add a zero to each number printed on this dial, the numbers represent frequency in kilocycles.





Radio station frequencies are usually expressed in kilocycles for convenience, so the term kilocycle (abbreviated kc.) is one which you should remember. (One kilocycle is equal to 1000 cycles.)

The tuning dial shown in Fig. 3 extends from 550 kilocycles to 2000 kilocycles, hence our superhet receiver covers the entire broadcast band and certain police bands just above the broadcast band.

Replacing Control Knobs. When a Radiotrician encounters a receiver By varying the resistances of R_s and R_4 (but keeping them equal), or by using different sizes of condensers (but keeping them equal also), we can change the frequency to which the bridge will be balanced. Thus, we can use the bridge to determine the fre-



Courtesy General Radio

FIG. 30. A typical laboratory type frequency meter using the Wien bridge.

quency of the source connected to it, provided we know the capacity of the condensers and the value of the resistors. A typical audio frequency bridge is shown in Fig. 30. The main dial varies the resistors, which are ganged together.

USING THE C. R. O. TO DETERMINE FREQUENCIES

From your previous study of the cathode ray oscilloscope, you know that you can get an image on the face of the tube which will be an exact reproduction of the applied voltage. Furthermore, when the sweep frequency is locked in with the incoming signal frequency, you'll get a single cycle of the incoming wave. Now, let's see how you can use these characteristics of the c.r.o. to measure frequencies.

As you know, a stationary pattern

can be obtained if the sweep frequency is a sub-multiple of the incoming frequency. The number of complete cycles in the pattern gives the relationship between the sweep frequency and the incoming frequency. For example, if the incoming wave is 1000 cycles and the sweep is set to 500 cycles, you will get a 2-cycle pattern as shown in Fig. 31A. Similarly, a wave four times the sweep frequency makes four complete cycles, as shown in Fig. 31B, and so on up to 15 or 20, depending on the size of the c.r.o. screen.

When the sweep frequency is adjusted the other way so it is above the incoming signal frequency, you no longer get an image of the incoming signal. Instead, there will be a series of more or less horizontal lines on the c.r.o. screen. When there are two stationary lines, the sweep frequency is twice the incoming signal; when there are three lines, the sweep is three times the frequency, etc. Fig. 31C shows five lines, so the sweep fre-



FIG. 31. The number of cycles or lines in the c.r.o. pattern can be used to indicate frequency.

quency is five times the incoming frequency.

Thus, if the sweep generator in the c.r.o. is calibrated, you can easily determine the approximate frequency of the incoming signal from the setting of the sweep controls and the pattern formed. You can calibrate the sweep by feeding known frequencies into the c.r.o. and noting the setting of the sweep frequency control necessary to produce stationary patterns.

LISSAJOUS FIGURES

Since the sweep of the c.r.o. may drift in frequency and thus be of varying accuracy, the calibrated sweep method is used only for rough measurements. When you want greater accuracy you should use Lissajous figures to measure frequencies with the c.r.o.

To use Lissajous figures, you must have an accurate source of known frequency. The unknown frequency is fed to the vertical plates of the c.r.o. tube, while the known standard is fed to the horizontal plates. An excellent source for the standard frequency is a.c. from a 60-cycle power line, which is accurately controlled by the power company.

If the frequency of the unknown and the known frequencies are exactly equal, you will get a single slanting line, an ellipse, or a single circle on the c.r.o. screen. The possible pattern



FIG. 32. Using imaginary lines A and B to determine the frequency of a Lissajous figure.

differences are due to phase shifts. However, any pattern which is a single line, oval or circle shows that the two a.c. voltages have the same frequencies.

On the other hand, if you get a



FIG. 33. Several typical Lissajous figures showing frequency ratios.

series of closed loops, the voltages have different frequencies. The position and number of the loops give the ratio of the applied frequencies; even fractional ratios can be determined in this way.

For example, suppose you get a pattern like that shown in Fig. 32A. To determine the frequency ratio, draw or imagine the lines A and B. Notice the number of times each line is touched by the pattern. Line A is touched four times, while line B is touched once. This means the frequency ratio is 4 to 1—that is, the signal applied to the vertical plates is four times that on the horizontal plates. Thus, if a 60-cycle voltage is applied to the horizontal plates, the vertical plate voltage has a frequency of 240 cycles.

If the image is turned around, as shown in Fig. 32B, so the figure touches the line A once and the line B four times, the ratio will be 1 to 4. This means the horizontal signal frequency will be four times the vertical signal frequency; if the horizontal voltage is still a 60-cycle signal, the vertical voltage must be a 15-cycle signal.

A number of typical patterns are shown in Fig. 33. In all cases, count-

ing the number of times the loops touch the imaginary lines A and B will pattern with a ratio of 5 to 3 shows give the frequency ratios. These ratios, multiplied by the frequency of the horizontal voltage, equal the frequency of the vertical voltage. Thus,

if the horizontal voltage is 60 cycles a that the frequency of the vertical voltage is $5/3 \times 60$, or 100 cycles. This method is quite accurate if the pattern is studied carefully.

THE N. R. I. COURSE PREPARES YOU TO BECOME A RADIOTRICIAN & TELETRICIAN (REGISTERED U.S. PATENT OFFICE) (REGISTERED U.S. PATENT OFFICE)

Lesson Questions

Be sure to number your Answer Sheet 30FR-1.

Place your Student Number on every Answer Sheet.

Send in your set of answers for this lesson immediately after you finish them, as instructed in the Study Schedule. This will give you the greatest possible benefit from our speedy personal grading service.

1. What factor is found when you divide the coil reactance by its resistance $Q = (X_L \div R)$? Q bactor of the Coul

- 2. When measuring the inductance of an iron-core coil, what value of polarizing (d.c.) current should flow through it?
- 3. When measuring the inductance of an iron-core coil which has a high resistance, would you use: 1, The reactance method; or 2, the three-voltage method, if the true inductance is to be found?
- 4. Which of the following condensers requires a polarizing voltage when checking the capacity: 1, Air dielectric; 2, paper; 3, ceramic; 4, electrolytic; 5, mica?
- 5. When using the resonance method of measuring inductance, what condition is indicated by a double-hump (two peaks) reading on the meter?

6. Using a resonant/circuit like the one in Fig. 10, what four factors of a coil can be measured or figured?

- Resistance, 9, inductance & Reactance 7. When finding the Q using the circuit of Fig. 11, what basic principle is being used in the measurement? Resonant voltage step-up.
- 8. Using the decade box of Fig. 13, to which contacts would you set switches SW_1 , SW_2 and SW_3 so that you would have 471 ohms between terminals A and B? SW_1 / SW_2 7 SW_3 4
- 9. When a bridge circuit is balanced, does the indicator show: 1, Maximum current; or 2, minimum current?
- 10. Suppose you are using a wavemeter, consisting of a coil and a condenser, which has no built-in indicator. If this wavemeter is brought near an oscillator tank circuit, what indication will be obtained on the oscillator plate current meter when the wavemeter is tuned to resonance?

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World Radio History

VARIED INTERESTS

Don't go stale! Keep a fresh, active interest in everything you do. When you work or when you play—work hard, play hard. When you study, concentrate on study. When you stop working, or playing, or studying—let go. Forget about it. The best vacation is merely a complete change from what you have been doing. Loafing is not a vacation—it is merely boredom.

There is nothing better for an office worker after hours than a brisk walk, a swim or a round of tennis. There is nothing better for an outdoor worker than a quiet hour with a book or a good newspaper, or listening to radio programs. But don't do the same things all the time. Vary your life as much as you can. Cultivate a keen interest in the world of which you are a part.

Keeping alert keeps you young. Keeping interested keeps up your energy. Together these things will make you able to learn more in a given period of time. Finally, the more you learn, the more easily you learn.

J. E. SMITH

TUNING INDICATORS AND AUTOMATIC FREQUENCY CONTROLS FOR RADIO RECEIVERS

31FR-1

NATIONAL RADIO INSTITUTE ESTABLISHED 1914 WASHINGTON, D. C.



Study Schedule No. 31

For each study step, read the assigned pages first atyour usual speed. Reread slowly one or more times. Finish with one quick reading to fix important facts firmly in mind, then answer the Lesson Questions for that step. Study each other step in this same way.

1. Making Receivers Easy to Tune; Current-Controlled Visual Tuning Indicators; Voltage-Controlled Visual Tuning Indicators - Pages 1-8

Here is a wealth of practical information which you'll appreciate even more once you begin working on receivers which have tuning indicators. Give particular attention to the section on cathode ray tuning indicators, because you will encounter these more than any other types. Answer Lesson Questions 1, 2 and 3.

2. Unique Tuning Aids; Essential Sections of an A.F.C. System; The Discriminator Circuit

Here we have brief but interesting descriptions of one-of-a-kind tuning indicators which are to be found on only a few models of receivers. One reading now will be enough, because you can always refer to this section later. Study thoroughly every bit of the information on a.f.c. systems, however, because this basic information applies also to f.m. receivers. Answer Lesson Questions 4, 5 and 6.

3. Review of Phase; Discriminator Circuit Action; The Oscillator Control Circuit

Study this material slowly because every word counts. It isn't exactly essential to understand all this theory in order to fix a receiver having an a.f.c. system or repair an f.m. receiver, but this knowledge will definitely help a lot in most jobs of this type. Furthermore, any man who understands how an a.f.c. system works is really deserving of promotion from the ranks of a beginning student to an advanced group. There are thousands of actual servicemen who do not understand the material you will study here.

4. Typical A.F.C. Circuits; Adjusting A.F.C. Systems - - Pages 19-28

Here are practical examples from actual receivers, to illustrate the theory you learned in the first part of this lesson. Take it slow and easy, because fast reading will get you nowhere in material of this nature. Answer Lesson Questions 7, 8, 9 and 10.

□ 5. Mail your Answers for this Lesson to N.R.I. for Grading.

6. Start Studying the Next Lesson.

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World Radio History

Tuning Indicators and Automatic Frequency Controls for Radio Receivers

Making Receivers Easy To Tune

THE tuning dial of a selective, modern receiver may require adjustments which are accurate to within one-hundredth of an inch movement of the dial pointer on its scale, if the desired signal is to be tuned in properly, free from distortion. The effects of improper tuning upon the wave form of a signal are clearly shown by the diagrams in Fig. 1.

Receiver designers recognize the difficulties involved in tuning a receiver with this high degree of accuracy, and for this reason have provided a number of tuning aids on the receiver. Large dial scales are provided, with correspondingly longer pointers, to make small movements of the pointer more readily distinguishable. Two-speed adjusting knobs. one turning the tuning condenser assembly directly and the other through a mechanical speed-reducing or vernier mechanism, further simplify accurate tuning.

A dial designed for accurate tuning is of little value if the listener has to rely upon his ears as a guide for proper tuning. The reasons for this are quite simple; loudness is not a dependable guide because about equal loudness is secured for several kilocycles on either side of resonance, due to A.V.C. action. Clarity of reception is likewise a poor guide, for the average individual is not capable of telling immediately whether or not distortion is present. (In a receiver having A.V.C. but no tuning indicator we could, of course, tune for maximum volume with minimum background noise, but few people realize that this can serve as a guide for correct tuning.)

Tuning Indicators. The introduction of automatic volume control practically forced radio designers to develop another improvement, the visual tuning indicator. Fortunately, there is in A.V.C. receivers a voltage or current which either reaches a maximum or minimum value when the receiver is correctly tuned. This woltage or current can be made to operate a simple meter-type indicator having either a needle pointer or a shadow indicator; other types of indicators used for this purpose include a cathode ray indicator tube (often



FIG. 1. These diagrams, copied from curves produced on the screen of a cathode ray oscilloscope in the N. R. I. laboratory, show what happens when a selective radio receiver is carelessly tuned. At Ais the audio output curve secured when the receiver is correctly tuned to a carrier modulated with a single pure sine wave; there is no distortion. The slightly-distorted audio output wave form produced when the receiver is tuned about 2 kc. above the input R.F. carrier frequency appears at B, while the severely distorted wave form at 10 kc. off-tune appears at C.

called a magic eye), a miniature electric lamp providing flash tuning, or an audio beat oscillator which provides an audible indication of incorrect tuning.

Visual tuning indicators are by no means a perfect solution to this problem of tuning, for even with the best types of indicators the average broadcast listener finds it difficult to tune exactly at all times. Correct tuning requires patience and careful watching of the tuning indicator; a person who is relaxing to enjoy a radio pro_x gram does not care to bother with the details of proper tuning. Radio engineers recognize that slight mistuning is relatively unimportant during the first few moments of listening. After the radio receiver has been in operation for an hour or so, however, even small amounts of distortion become objectionable although they cannot consciously be recognized by the listener. Distortion appears to affect the nervous system, making a person tired of listening even to a good program.

Since the success of the entire Radio industry, including broadcasters and receiver manufacturers alike, depends upon having a large listening public, it is only natural that radio receiver engineers continued their search for devices and methods which would simplify the tuning of receivers. Their goal was a system which would permit even a small child to tune in the receiver just as well as could a radio expert.

Automatic Frequency Controls. As you know, in a superheterodyne receiver it is the local oscillator frequency which, more than anything else, determines the value of the I.F. signal. When this I.F. signal frequency is different from the I.F. value of the receiver, due to improper tuning, distortion will be present and sensitivity will be poor.

The superheterodyne receiver circuit is well suited for automatic tuning; even though it may be as much as 5 kc. off resonance, the tuning can be corrected merely by adjusting the oscillator trimmer condenser or by changing the inductance of the oscillator coil. \checkmark

With this basic superheterodyne principle in mind, radio engineers produced a circuit for developing automatically a voltage which is proportional to the amount by which the oscillator is off resonance. This voltage is fed into a special vacuum tube circuit which is connected to the oscillator and has the peculiar ability of being able to change the inductance or capacity of the oscillator circuit just enough to produce the correct I.F. signal value. This system of automatic frequency control is commonly called A.F.C.

When a receiver is equipped with A.F.C., the listener merely tunes in the desired signal as best he can without any particular attention to accuracy of tuning; the A.F.C. system then automatically completes the tuning procedure. V

Automatic frequency control sys-



Fig. 2. Examples of current-controlled tuning indicators using an inexpensive milliammeter movement which consists essentially of a solenoid (coil) of about 1000 ohms resistance, inside which moves a long, pointed steel vane to which is attached the indicating pointer. The greater the D.C. current through the solenoid, the greater is the force which pulls the vane into the solenoid and moves the pointer. At C is a shadowgraph tuning indicator using a milliammeter movement; its operating principle is illustrated in Fig. 3.

tems are used chiefly in those larger receivers which have some type of automatic push button tuning system; in fact, A.F.C. is essential in a receiver having the electro-mechanical (motor-driven) type of automatic tuning system in order to correct for slight inaccuracies in the mechanical action of the tuning mechanism and compensate automatically for oscillator frequency drift.

Tuning aids and automatic frequency control systems provide additional problems for the Radiotrician. The operating principles of the devices in general use for tuning aids must be thoroughly understood before service can be rendered, so these principles will now be taken up.

Current-Controlled Visual Tuning Indicators

Meter-Type Indicator. Meter-type tuning indicators are ordinarily connected into the plate supply lead going to one or more A.V.C.-controlled stages. The average D.C. plate current drawn by a screen grid or pentode tube with normal C bias is about 5 milliamperes; this plate current drops practically to zero when the A.V.C. system increases this negative C bias to 40 or 50 volts. A 0-5 ma. milliammeter movement can therefore be used as a meter-type tuning indicator.

In Fig. 2A is shown one form of a meter-type tuning indicator. This unit is attached to the receiver chassis in such a way that the face of the meter can be viewed through a hole in the receiver panel. An escutcheon plate like that shown in Fig. 2B is often used with this meter to improve its appearance, and a pilot lamp is sometimes mounted below the meter to illuminate the scale through the window which can be seen in Fig. 2A. The no-current position of the pointer is at the extreme right, as shown in these views. With the meter connected to indicate D.C. plate currents of an A.V.C.-controlled stage, the current through the meter is a maximum when no signal is tuned in, and the pointer will be at the extreme left. Tuning in a carrier signal causes A.V.C. action to decrease the D.C. plate current for the A.V.C.-controlled stages, and the tuning meter pointer therefore moves toward its zero position at the extreme right. The listener tunes the receiver for a maximum deflection to the right, or in other words,

tunes for greatest swing to the right, for this corresponds to at-resonance conditions in the receiver circuit.

There are a number of variations of the simple meter movement just described, but all depend upon the same operating principles. In one case a small, round black disc is mounted on the end of the meter pointer; this disc moves over a series of concentric circles, alternately black and white, like those on rifle targets. The no-current



FIG. 3. In this shadowgraph tuning indicator, a large current causes the meter movement to place the vane in position B, giving a maximum-width shadow. Tuning in a station reduces the current, making the vane take various positions, such as at C and D; the listener tunes for minimum shadow, since this corresponds to exact resonance for any station.

position of the disc is at the center of the target, and the listener therefore tunes to get this pointer' as close as possible to the center; manufacturers' instructions say: "Tune for a bull's eye."

Shadowgraph Tuning Indicator. Perhaps the most widely used variation of the tuning meter movement is the shadowgraph tuning indicator shown in Fig. 2C. This contains a frosted glass or celluloid screen which is mounted so as to be visible from the front of the receiver panel. Behind this screen is a pilot lamp, the light from which passes through a small window and past a small rectangular vane attached to the meter movement before reaching the screen. When meter current is a maximum. this vane blocks most of the light, casting a shadow on the screen. As resonance is approached when tuning in a station, the meter movement ro-



FIG. 4. Loosening the set screw of a receiver control knob with a small screwdriver. Some knobs may have two set screws.

having a broken or missing control knob, he will first try to obtain a new knob exactly like those already on the set.

If an exact replacement knob is not available, he will secure an entire set of new knobs, having a design and color which will harmonize with the finish of the receiver cabinet.

Removing the Chassis

In order to get a good look at the radio parts in our receiver, we will have to remove the *chassis* from the cabinet. The chassis includes the metal *chassis base* and all parts mounted on this base.

Removing Control Knobs. We first loosen the tiny set screw in each control knob, using a small screw-driver as illustrated in Fig. 4, then remove the knobs.

Sometimes knobs are held in position by stiff flat springs inside the knobs, instead of screws. These knobs are removed by pulling firmly. If it is difficult to get a good grip on a small knob, slip a handkerchief between the knob and the cabinet, and pull on the ends of the handkerchief.

Removing Chassis Screws. Next, we use a medium-size screwdriver to loosen the two wood screws which fasten the chassis base to the bottom of the receiver cabinet, as illustrated in Fig. 5.

In some receivers, the chassis is fastened with bolts which pass through the bottom of the cabinet. These are removed either with a screwdriver or socket wrench, depending upon the type of bolt head employed, after turning the receiver on its side so





that the bottom of the cabinet is accessible.

The chassis-removing procedure just described applies to practically any radio receiver, so let's repeat it here as a practical radio servicing rule: To remove the chassis from the cabinet of a radio receiver, first take off the control knobs, then remove the chassis hold-down screws or bolts. If the chassis will not come out yet, look for additional hold-down screws around the dial and loudspeaker. tates the vane towards its zero position, giving a minimum-width shadow. Details of this arrangement are shown in Fig. 3A, and the effects of vane position upon the width of the shadow are shown in Figs. 3B, 3C and 3D.

Circuit Connections. When a tuning meter is to be actuated by the D.C. plate current of only one A.V.C.controlled stage, connections into the plate supply lead of this A.V.C.-controlled tube may be made as shown



FIG. 4. Connections for a meter-type tuning indicator for a single A.V.C.-controlled stage (A) and for the common plate supply lead to two A.V.C.controlled stages (B).

in Fig. 4A. Notice that by-pass condenser C_{P1} prevents R.F. and I.F. currents from passing through the tuning indicator. The resistance of the tuning indicator drops the D.C. plate voltage about 5 to 10 volts.

When the D.C. plate current to more than one A.V.C.-controlled amplifier stage is passed through the tuning indicator, connections into the plate supply lead going to these A.V.C.-controlled tubes may be as shown in Fig. 4B. Naturally the current range for the indicator in this circuit must be greater than for the tuning indicator in Fig. 4A. The meter movement should be selected to give nearly full-scale deflection to the left for a no-carrier-signal condition, in order to give maximum meter movement when a station is tuned in. Notice that by-pass condensers C_{P1} and C_{P2} are used in this arrangement to keep R.F. and I.F. currents out of the tuning indicator.

Voltage-Controlled Visual Tuning Indicators

Cathode Ray Tuning Indicator. A special cathode ray tube which acts essentially as a high-resistance voltmeter is used in many receivers as a voltage-controlled visual tuning indicator. The D.C. voltage developed across the load resistor of a diode second detector-A.V.C. tube in a receiver is used to control the cathode ray tuning indicator, since this voltage varies with carrier level.

The operating principle of a cathode ray tuning indicator tube is essentially the same as that of the cathode ray oscilloscope tube. Electrons striking a surface coated with a fluorescent material (such as willemite) cause the surface to glow with a greenish light. The diode detector load voltage determines the amount of surface which glows at any time, and thus we have a simple and highly attractive tuning indicator.

In a cathode ray tuning indicator an oxide-coated cathode, heated by a filament, is mounted in the center of a cone-shaped anode having a coating of fluorescent material on its inner surface, as indicated in Fig. δA . When a D.C. voltage is applied between the cathode and the anode, the electrons emitted by the heated cathode will strike the inner surface of the anode, called the *target*, and the entire inside of the circular cone or target will glow with a greenish light. \checkmark In addition, a thin, vertical dart or vane is located between the target and the cathode, as shown in Fig. 5B. This dart is known as the control



FIG. 5. Essential elements of a cathode ray tuning indicator tube.

electrode, for it determines how much of the target will glow.

When the control electrode is connected to the cathode, it has no electrical influence upon the electrons emitted by the cathode, but does have a physical blocking effect which prevents electrons from reaching that portion of the target directly behind it. We thus have a shadow in back of the control electrode, as indicated in Fig. 6A. Making the control electrode negative with respect to the cathode, as in Fig. 6B, widens this shadow, for the control electrode now has a repelling effect on the emitted electrons. When the control electrode is made positive with respect to the cathode, however, it serves to speed up the emitted electrons and to make them bend towards the shadow region. narrowing the shadow as indicated in



FIG. 6. Variation of shadow angle with control electrode potential in a typical cathode ray tuning indicator tube.

Fig. 6C. When the control electrode is highly positive, as in Fig. 6D, the shadow becomes very narrow or disappears entirely.

When a receiver having a diode de-

tector is tuned to resonance, the diode load voltage increases in a negative direction with respect to the chassis or ground. If we want the shadow angle to close up as a station is tuned in, we must convert this increasing negative voltage into an increasing positive voltage for the control electrode of the cathode ray indicator



Courtesy National Union Radio Corp. FIG. 7. Cutaway view showing construction of 6G5 cathode ray tuning indicator tube.

tube. This can quite easily be done by introducing a triode amplifier between the detector load and the control electrode of the indicator tube; in an actual cathode ray indicator tube like that shown in Fig. 7, this triode amplifier is built into the same envelope as the target system.

The schematic circuit diagram in Fig. 8 gives in simplified form exactly the same information as the pictorial sketch in Fig. 7. In addition, this

diagram shows that the target connects directly to the B++ D.C. supply source, while the plate of the triode tube connects to this source through resistor R, which is usually 1 megohm. The common cathode for the two tubes is grounded to the receiver chassis, and the A.V.C. voltage is applied between the control grid of the triode section and the cathode.

When a receiver which has a cathode ray tuning indicator is tuned be-



FIG. 8. Schematic symbol for a cathode ray tuning indicator tube. External resistor R is always used between the triode plate and the target. Circuit variations have little effect upon target current, since the target receives a constant D.C. supply voltage.

tween stations, there is no carrier signal and consequently there is no A.V.C. voltage developed across the Under this detector load resistor. condition the C bias for the triode section of the indicator tube is zero $(E_{AVC} \text{ is zero})$ and D.C. plate current $I_{\rm P}$ for the triode section depends upon the B++ supply voltage, the value of resistor R and upon the plate-tocathode resistance of the triode section. Since this plate-to-cathode resistance is quite low for zero grid bias, current I_P is high, and creates a high voltage drop across R. The triode plate voltage is consequently considerably less than the target voltage (because of this drop across R). The control electrode (connected directly to the triode plate) likewise has considerably less voltage than the target when no station is tuned in, and the shadow angle is therefore large.

When the receiver is tuned to a desired station carrier signal (is tuned to resonance), the A.V.C. voltage developed across the detector load increases, driving the triode grid of the indicator tube negative and increasing the D.C. plate resistance of the triode section. The result is reduced D.C. plate current and a lowered voltage drop across resistor R. The triode plate voltage and the control electrode voltage become more nearly equal to the target voltage, and the increasing positive voltage on the control electrode decreases the shadow angle on the target. This shadow angle will be a minimum when the receiver is tuned exactly to resonance for a particular station. For distant stations the minimum shadow angle will be considerably greater than for local stations.

It is possible to adjust a cathode ray indicator tube circuit to make the shadow vanish for either small or large A.V.C. voltages. With circuits designed so the shadow closes on low A.V.C. voltages, however, the tuning in of strong carrier signals will completely close the shadow before resonance occurs and an exact indication of resonance will not be possible. On the other hand, if the circuit is designed so the shadow will completely close only for strong local carrier sig-√ nals, then the change in shadow angle for weak signals may be inadequate for accurate tuning. (This is not a serious drawback, however, for it is ordinarily easy to tune in a weak station accurately by ear.)

The A.V.C. voltage is, of course, the primary control upon the shadow angle. The angle for zero A.V.C. voltage is governed essentially by the design of the tube, but the amount which the shadow closes for a given increase in A.V.C. voltage is determined by the target voltage and by the value of series resistor R.

Curve 1 in Fig. 9 tells how the shadow angle varies with A.V.C. voltage when the supply voltage is low (100 volts) and the ohmic value of R is low (.5 megohm). Observe that the shadow angle is 90 degrees for zero A.V.C. voltage, and the shadow is completely closed (0° shadow an-



FIG. 9. This graph shows how shadow angle varies with grid voltage for a 6G5 cathode ray tuning indicator tube. Curve 1: R = .5 megohm and B++voltage = 100 volts; curve 2: R = 1 megohm and B++ voltage = 250 volts.

gle) at an A.V.C. voltage of about -8 volts (the curves in Fig. 9 apply to a 6G5 tube). Curve 2 in Fig. 9 is for a high plate supply voltage and a high value of R; in this case the shadow does not close until the A.V.C. voltage has driven the triode tube grid to -23 volts.

We can draw the following conclusions from the curves in Fig. 9: 1. to make the shadow close at a more negative grid bias value, increase the supply voltage to the tube: 2, to make the shadow close at a less negative value (so it will close for weak carrier signals), reduce the supply voltage to the tube. To secure the desired open shadow angle, usually about 90 degrees, adjust resistor R when no carrier signal is present. With these facts in mind, you will be able to adjust a cathode ray indicator tube for satisfactory operation on either local or distant stations. Most receivers are adjusted for a characteristic curve approximating curve 2 in Fig. 9, for this gives a reasonable change in shadow angle at all reasonable carrier levels.

Adding a Cathode Ray Indicator Tube to a Receiver. A suitable circuit is shown in Fig. 10A; all parts which are added to the original receiver circuit are shown in heavy lines. Note that the grid of the 6G5 cathode ray tuning indicator tube is connected through a 1-megohm resistor R_1 to the A.V.C. voltage source in the receiver. This connection could be to the minus terminal of the diode detector load, but since this terminal may furnish as much as -40 volts when a strong local station is tuned in, it is better to place voltage divider R_2 - R_3 across the diode load and make a connection to this.

The larger the value of R_2 with respect to that of R_3 , the lower will be the maximum A.V.C. voltage applied to the indicator tube. R_1 prevents the 6G5 tube from shorting the diode load



FIG. 10A. This diagram shows you how a cathode ray indicator tube would be connected between the A.V.C. supply terminal and ground in a receiver. The only extra parts required, in addition to the type 6G5 indicator tube and its standard 6-prong socket, are resistors R, R_1, R_2 and R_3 , and condenser C_1 . You will sometimes find the control electrode and target represented as shown here rather than as in Fig. 8, so learn to tecognize both types of diagrams.

or the voltage divider; in addition, R_1 serves with C_1 as an A.F. filter which prevents the shadow from flickering at an A.F. value.

Wide-Angle Tuning. The shadow angle for a no-signal condition in the

average cathode ray tuning indicator tube is 90°, but this can be increased approximately to 180° by inserting an ordinary triode amplifier tube between the A.V.C. supply terminal and the indicator tube in the manner shown in Fig. 10B.

When there is no carrier signal in the diode second detector circuit of a receiver, there is no A.V.C. voltage developed across the diode load, and consequently the grid of VT_1 in Fig. 10B will be at zero bias. The plate



FIG. 10B. A shadow angle of approximately 180° can be secured with this cathode ray tuning indicator circuit.

current of this tube is then high, and the plate-cathode resistance of VT_1 drops so low that point x is nearly at ground potential. Since the cathode of VT_2 is +125 volts with respect to ground, the triode plate and the control electrode of VT_2 are -125 volts with respect to the cathode of this tube; under this condition the shadow angle is approximately 180°.

When a highly negative A.V.C. voltage is applied to the grid of VT_1 (tuning the receiver to a powerful local station would do this), the plate current of VT_1 drops nearly to zero; under this condition the D.C. resistance of VT_1 is so much higher than the 1-megohm resistance of R that point x practically assumes B+ potential of +250 volts. This places the triode plate and the control electrode at about +125 volts with respect to the cathode of VT_2 , giving zero shadow angle. Weaker signals give shadow angles between 0 and 180°, with the change in shadow angle while tuning in a station being about twice as great as for an ordinary indicator circuit; this, of course, makes more accurate tuning possible.

Unique Tuning Aids

Although the milliammeter and the cathode ray tuning indicator tube are the most widely used tuning aids, a number of other unique methods have been developed. Three of these will be considered at this time: 1, the G.E. Colorama indicator, using red and green lamps; 2, the neon lamp flasher; 3, the zero beat tuning indicator. \checkmark

G. E. Colorama Tuning Indicator. The circuit diagram of the Colorama tuning indicator used in a number of General Electric receivers is shown in Fig. 11. The A.V.C. voltage developed by the diode detector in the receiver is applied to the control grid of an ordinary type 6C5 triode tube (designated as the Colorama tuning indicator tube). In the plate circuit of this tube is the primary of the Colorama tuning reactor (an iron-core transformer) while across the secondary of this transformer is an arrangement of three green lamps in series and four red lamps in series-parallel, connected as indicated in Fig. 11. Condenser C52, across the primary of the transformer, serves to filter out any A.F. components in the A.V.C. voltage; R28 and C53 are power pack. supply filters.

When the A.V.C. voltage is low or zero, as it is when the receiver is tuned off resonance or between stations, the C bias on the 6C5 tube is zero and D.C. plate current for this tube is therefore quite high. This high plate current flowing through the primary of the Colorama tuning reactor saturates its iron core. This core saturation lowers the reactance of the secondary winding, so that it is practically a short-circuit path for 60-cycle alternating current. Under this condition the green lamps are in effect shunted by the 15-ohm resistor R30, and the current which is sent through the bank of red lamps and R30 by secondary winding S on the receiver power transformer is high enough to light all of the red lamps. We thus have the red lamps glowing whenever the A.V.C. voltage is reduced by funing between stations.

Now let us see what happens when a station is tuned in. Tuning the receiver to resonance increases the negative A.V.C. voltage applied to the grid of the 6C5 tube, and the plate current of this tube drops. The core of the Colorama tuning reactor becomes less saturated as primary current drops, and consequently the inductance of secondary winding L28 increases. The reactance of this secondary winding at 60 cycles is correspondingly increased, reducing the shunting effect of R30 and thus increasing the resistance of the entire circuit across transformer winding S. Less current now flows through the red lamps, and they begin to grow dim.

At a critical negative A.V.C. voltage, full current (.150 ampere in this case) flows through the green lamps, lighting them brightly, while the current through a red lamp is only half of the circuit current, because of the series-parallel connection. Resistor R30 controls the maximum current which can flow through the red lamps when the receiver is off-tune, while resistor R29 controls the current which the green lamps can draw when the receiver is tuned to a station, and also controls to a certain extent the current for the red lamps. These resistors must be properly adjusted if the red lights are to illuminate the station selector dial when the receiver is off resonance, and the green lights are to illuminate the dial when the receiver is at resonance.

Neon Lamp Flasher Indicator. A single neon lamp connected into a circuit which causes it to glow only when a station is tuned in, serves as the tuning indicator for a number of Silvertone receivers. The circuit diagram is shown in Fig. 12; an ordinary I.F. transformer with weak coupling



FIG. 11. Circuit diagram of the Colorama tuning indicator used in General Electric Models E-101, E-105 and E-106 receivers. The green lamps light up when the receiver is properly tuned; red lamps illuminate tuning dial whenever receiver is of tune. The reactance values given here for the primary and secondary (L27 and L28) of the Colorama tuning reactor are for 60 cycles A.C. Ohmmeter measurements will therefore give considerably lower values.

between the tuned primary and tuned secondary is connected between the primary of the last I.F. transformer and the 6B7 duo-diode-pentode which is designated as the flasher tube. Primary circuit L_1 - C_1 of this extra I.F. transformer forms a series resonant circuit which is connected across condenser C in the last I.F. transformer circuit, with the result that the small I.F. voltage drop across condenser Cis applied to L_1 - C_1 . At resonance this series resonant circuit acts as a low resistance, and hence does not affect the plate load of the last I.F. stage. Loose coupling between L_1 and L_2 insures higher selectivity for resonant circuits L_1 - C_1 and L_2 - C_2 than could be secured in the I.F. amplifier, for we are not concerned with fidelity in the tuning indicator circuit.

The I.F. voltage developed across resonant circuit L_2 - C_2 is rectified by the diode section of the 6B7 flasher tube, and a rectified voltage is developed across the 1-megohm diode load resistor. The control grid of the pentode section is connected to the negative terminal of this load resistor.



FIG. 12. Neon lamp flasher indicator circuit as used in Silvertone Models 1722 and 1732 receivers. The neon flasher lamp glows only when the receiver is tuned correctly.

The plate of the pentode section receives its D.C. voltage through 130,-000-ohm resistor R; between the plate and the cathode is the flasher neon lamp which glows only when its terminal voltage exceeds a definite value. Furthermore, the terminal voltage of this neon lamp is equal to the supply voltage minus the voltage drop in resistor R.

When the receiver is tuned off resonance, little or no I.F. voltage is applied to the diode section of the 6B7 flasher tube. As a result, no rectified voltage exists across the 1-megohm resistor, the C bias on the pentode section is practically zero, and plate current of the pentode section is high. This plate current produces a large voltage drop across resistor R, making the neon lamp voltage too low for it to glow. Since resonant circuits L_1 - C_1 and L_2 - C_2 are highly selective, the receiver must be tuned exactly to resonance before enough I.F. signal can get through these circuits to excite the diode section and produce sufficient negative voltage across the 1megohn resistor to reduce the pentode plate current to a low value. Reducing the plate current reduces the voltage drop across R and raises the neon_ lamp voltage. The lamp therefore glows when the receiver is properly tuned to a station. When the receiver is tuned rapidly from station to station, the lamp flickers each time a station is passed; this is why this particular arrangement is known as a flasher indicator.

Zero Beat Tuning Indicator. If the I.F. carrier signal produced by the mixer-first detector of a receiver is mixed with a locally-produced signal having exactly the correct I.F. value, and this mixing occurs just ahead of the second detector, an audio beat (low-frequency audio note) will be heard in the loudspeaker whenever the receiver is improperly tuned to a station.

When the receiver is properly tuned, the difference between the two I.F. signals will be less than 30 cycles, and the beat note will therefore be inaudible. When the receiver is tuned off a station, there will be no incoming I.F. signal to beat with the locally produced signal, and consequently no audio beat will be heard. A zero beat indicator of this type is entirely satisfactory provided the listener does not object to the squeal which occurs while tuning in a station. Furthermore, this indicator is effective regardless of whether the I.F. amplifier is peaked or band-passed. A zero beat tuning indicator is also of great value when tuning in continuous wave code signals, for it provides an audible tone for these signals.

The special local oscillator circuit which is used as a zero beat tuning indicator in one receiver is shown in Fig. 13. The oscillator tank circuit is made up of L_1 and C_1 , with C_1 being adjusted to make this circuit produce a frequency exactly equal to the I.F. value of the receiver. A.C. plate



FIG. 13. Circuit of the R.F. oscillator which serves as a zero beat tuning indicator in the Grunow Model 12A-1241 receiver.

current flowing through coil L_2 induces in L_1 the feed-back voltage required for oscillation. The A.C. plate current produced by this oscillator flows through a 2,000-ohm resistor which is also in the plate supply lead of the last I.F. amplifier stage. The result is that an A.C. voltage is superimposed on the D.C. plate voltage applied to this last I.F. stage. Two signals, the exact I.F. signal produced by the oscillator and the I.F. signal produced by tuning the receiver to a station, reach the second detector. After detection, we have left the desired audio signal and a beat note equal to the difference between the two I.F. signals; when the receiver is accurately tuned to a station, this difference will be zero, the beat note will disappear, and we have what is known as zero beat.

Essential Sections of an A.F.C. System

The two essential sections of an A.F.C. system are: 1, the discriminator, which usually produces a positive D.C. control voltage for below-normal I.F. signal frequencies, and a negative D.C. control voltage for abovenormal I.F. signal frequencies with the value of this voltage being proportional to the amount of error in tuning; 2, the oscillator control circuit, which converts the D.C. control voltages (produced by the discriminator) into changes in the effective inductance shunting the oscillator coil. thereby compensating for errors in tuning. The positions of these sections with respect to the other stages of a conventional superheterodyne receiver having A.F.C. are clearly shown by the box diagram in Fig. 14. Note that the I.F. amplifier feeds both the discriminator and the second detector.

The Discriminator Circuit

Typical Discriminator Circuit. The easiest way to see how the discriminator circuit in an A.F.C. system can furnish a D.C. control voltage of suitable polarity is to consider an actual discriminator circuit, as given in Fig. 15 We will first "get our bearings" by analyzing this circuit in a general manner, after which we will be ready to consider in detail the manner in which the D.C. control voltage is produced.

In the input circuit for the discriminator tube in Fig. 15 we find parallel resonant circuit $L_{\rm P}-C_{\rm P}$ serving as the plate load for the last I.F. stage, and inductively coupled to series resonant



FIG. 14. The two sections shown in dotted lines must be added to a superheterodyne receiver in order to secure automatic frequency control.

circuit L_1 - L_2 -C. Both of these resonant circuits are tuned to the exact I.F. value of the receiver. The coil in the series resonant circuit is split into two section of equal size L_1 and L_2 , and

the plate (high R.F.) terminal of the parallel resonant circuit is connected to the common terminal of these coils through D.C. blocking condenser $C_{\rm B}$.

The discriminator tube is a double diode rectifier tube (a 6H6 tube is most generally used), with the plate of each diode section connected to one terminal of the series resonant circuit and the cathodes connected together through equal-value resistors R_1 and R_2 . Each resistor is shunted by a bypass condenser (C_1 and C_2) and one cathode is grounded directly, hence age which we will designate as e_8 . This voltage acts in series with L_1 , L_2 and C in our series resonant circuit, causing a current to flow through the two coil sections. This current develops I.F. voltage e_1 across coil section L_1 and develops I.F. voltage e_2 across coil section L_2 . For any condition of tuning, voltage e_1 will always be equal to e_2 in magnitude.

We can now see that I.F. voltage e_1 acts in series with I.F. voltage e_P on diode section D_1 , and the resulting rectified electron current i_1 flows



FIG. 15. Conventional discriminator circuit.

both cathodes of the discriminator tube are at ground I.F. potential.

Notice now that the I.F. voltage developed across the parallel resonant circuit in the last I.F. stage is applied across choke coil CH, this I.F. voltage being designated as $e_{\mathbf{P}}$ (condenser $C_{\mathbf{B}}$ provides a direct path for I.F. current from the plate end of the parallel resonant circuit to point 1 and one terminal of the choke coil, and the path from the other end of the parallel resonant circuit is through by-pass condenser C_3 to the receiver chassis, through the chassis to the grounded end of R_2 and then through by-pass condenser C_2 to point 2 and the other terminal of the choke coil). This I.F. voltage $e_{\mathbf{P}}$ developed across choke coil CH is in turn applied to points 1 and 2 in the discriminator circuit.

Current *i*, flowing through the coil of the parallel resonant circuit, induces into secondary coil L_1 - L_2 a voltthrough R_1 , developing across this resistor a D.C. voltage having the polarity indicated in Fig. 15. Likewise, e_2 and e_P act upon diode section D_2 , and rectified electron current i_2 produces a D.C. voltage across R_2 with the polarity shown.

Since we are dealing with A.C. voltages in this discriminator circuit, phase must of course be taken into account when we consider their combined effects.

Correct Tuning. When the phase difference between e_1 and e_P is the same as that between e_2 and e_P , the net voltages acting upon diode sections D_1 and D_2 will be equal in magnitude; equal values of rectified current will then flow through the two resistors, making the D.C. voltage drop across R_1 equal to that across R_2 . The net D.C. control voltage produced across these two resistors (between point δ and ground) will be zero since these voltage drops are of opposite polarity, and the A.F.C. system will have no effect upon the oscillator. This is, of course, the condition for correct tuning of the receiver to a station.

Incorrect Tuning. When the frequency of the received I.F. signal does not correspond to the resonant frequency of the tuned circuits in the



FIG. 16. Vector diagrams showing conditions in the discrimioator circuit when the receiver is correctly tuned.

discriminator (because of incorrect tuning), e_1 and e_2 no longer have the same phase relationship with e_P , and consequently the A.C. voltages acting upon the diode sections are different in magnitude. Unequal rectified currents through R_1 and R_2 produce unequal voltage drops which, when combined, leave the desired D.C. control voltage to act upon the oscillator control circuit and correct for the error in tuning.

Since any detailed analysis of the action of an A.F.C. system must necessarily involve phase, it will be well at this time to review the fundamental facts about phase which you have already studied.

Review of Phase

All the information you need know about the phase relationships of voltages and currents for the parts in an A.F.C. circuit is given in Chart 1, page 27. The essential facts presented by this chart, particularly by the vector diagrams in the fourth column, are:

Resistors: The voltage across a resistor is in phase with the current through it.

- Coils: The voltage across a perfect coil leads the current through it by 90° (the current therefore lags the voltage by 90°).
- Condensers: The voltage across a perfect condenser lags the current through it by 90° (the current therefore leads the voltage by 90°).
- Transformers: The voltage *induced* in the secondary winding of a transformer (the open-circuit secondary voltage) is 180° out of phase with the primary voltage.

Discriminator Circuit Action

When the I.F. Signal Frequency is *Correct.* We will first analyze the discriminator circuit in Fig. 15 for the condition where the input I.F. signal frequency exactly corresponds to the I.F. value of the receiver. In this case parallel resonant circuit $C_{P}-L_{P}$ and series resonant circuit L_1 - L_2 -C will be resonant to the incoming signal. To determine the net A.C. voltages acting on diode section D_1 , we must add together A.C. voltages $e_{\rm P}$ and e_1 vectorially, so as to take into account the phase relationship of these two voltages. Likewise we must add together e_2 and e_p vectorially to find the net A.C. voltage acting upon diode section D_2 . First of all, we must choose some voltage or current for reference purposes; since $e_{\mathbf{P}}$ is common to all circuits under study, we shall use it as our reference voltage. To fix this fact in mind, we draw our vector $e_{\mathbf{P}}$ along the reference line in our vector diagram, as in Fig. 16A, using any convenient scale to determine its length.

The voltage e_8 which is induced in the secondary of the discriminator transformer is 180° out of phase with the voltage across the primary (see diagram *P* in Chart 1), and consequently we can say that e_8 is 180° out of phase with e_P . We therefore place vector e_8 on our vector diagram

Radio Receiver Tubes

Now we can slide the chassis out from the back of its cabinet. When this is done, all of the parts on top of the chassis base are clearly visible, as shown in Fig. 6. Study this view for a few minutes, to become familiar with the positions of the tubes and the labeled large parts.

Before you can understand how parts work together in radio circuits, and before you try to locate defective parts, you should have a general idea of how these radio parts look, how they are built, and how they act on signals and circuit currents when operating properly. Therefore, we concentrate upon radio parts in this lesson, starting with tubes.

The five tubes in our superheterodyne radio receiver are all on top of the chassis base, and are easily recognized even though one of them is almost completely enclosed in a metal shield.

Tube Shields. First of all, we'll take off that metal shield and see what the hidden tube looks like. We remove the clip from the top cap of the tube, then pull the shield straight up with one hand as shown in Fig. 7.

Yes, there's just an ordinary radio tube inside this shield, a type 6K7 tube serving in the i.f. amplifier stage. It's an important tube, for it receives extremely weak desired signals and boosts them a tremendous amount. But undesired signals would get this same amplification if they got into this tube circuit, so we must use the metal *shield* to protect the tube elec-



FIG. 6. Rear view of our superheterodyne receiver, with the chassis almost out of the cabinet.
in the opposite direction from $e_{\mathbf{P}}$, as indicated in Fig. 16A.

Since the frequency of induced secondary voltage e_s is exactly the same as the resonant frequency of series resonant circuit L_1-L_2-C , this circuit acts like a resistance at resonance and e_8 will send through the circuit a current i_8 which is in phase with e_8 , as shown in Fig. 16A. Current i_8 flows through L_1 and L_2 , developing across each of these coil sections an A.C. voltage which leads the current by 90° (see diagram H in Chart 1).

Rather than confuse the diagram in Fig. 16A by adding more vectors to it, let us redraw it in Fig. 16B and place vectors e_1 and e_2 on this new diagram. Before these vectors can be drawn, however, one other factor must be taken into account. For any direction of current flow i_8 through L_1 and L_2 , one of the coil voltage drops will be opposite in polarity (180° out of phase) with the other coil voltage drop in so far as $e_{\mathbf{P}}$ is concerned. This is because $e_{\mathbf{P}}$ acts in opposite directions through the two coils (it acts in the direction from point 1 to point 3through L_1 and from point 1 to point 4 through L_2). For this reason, if we show voltage drop e_1 as leading i_8 by 90°, we must show e_2 as lagging i_8 by 90°, making e1 180° out of phase with e_2 . (We could just as well make e_2 lead i_{s} by 90° and make e_{1} lag i_{s} by 90°, for the same results would be secured.) Now we can draw in vector e_1 and e_2 , as shown in Fig. 16B.

The next step is to find the net A.C. voltage acting upon diode section D_1 through resistor R_1 . As was said before, this voltage will be equal to $e_P + e_1$ with phase taken into account. We can add these two voltages quite easily on the vector diagram in Fig. 16B; we simply complete the parallelogram (rectangle) of which e_1 and e_P form two sides, as indicated by the dotted lines x and y in Fig. 16B, and then draw a line from the center of our vector diagram to the intersection of these dotted lines. This line now represents the vectorial sum of voltages e_P and e_1 , so we label this vector in this way. The length of vector $e_1 + e_P$ corresponds to the magnitude of the A.C. voltage acting upon diode section D_1 . In a similar manner we determine that vector $! e_2 + e_P$ is the net A.C. voltage acting upon diode section D_2 .

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These net A.C. voltages acting upon the diode sections will always be exactly equal when the I.F. signal frequency corresponds exactly to the I.F. value of the receiver. The diode sections D_1 and D_2 will consequently pass currents of equal value, and the D.C. voltage developed across R_1 by diode current i_1 will exactly equal the D.C. voltage developed across R_2 by diode current i_2 . Note that these currents flow in opposite directions through R_1 and R_2 , making the voltage drops across these resistors have the polarities indicated. The net D.C. voltage developed across the two resistors (between point 5 and ground) for the oscillator control circuit is therefore zero whenever the I.F. signal is of the exact value.

When the I.F. Signal Frequency is High. When the I.F. signal frequency entering the discriminator circuit is higher than the I.F. value of the receiver (higher than the resonant frequencies of $L_{P}-C_{P}$ and $L_{1}-L_{2}-C$), we will naturally expect the discriminator circuit to produce a D.C. control voltage for the oscillator control circuit. Let us see how this is done.

Primary voltage e_P will again be used as our reference voltage for the vector diagram. The induced voltage e_S will again be 180° out of phase with the primary voltage e_P , so we draw the vectors for these two values as in Fig. 17A.

Since the frequency of e_8 is higher

than the resonant frequency of series resonant circuit L_1 - L_2 -C, the reactance of coils L_1 - L_2 will be greater than the reactance of condenser C, and the entire series resonant circuit will act as an inductance whose reactance is equal to the difference between the above-mentioned reactances. We thus have voltage es acting upon a coil; if there were no resistance in this circuit, we could say that circuit current is lags the applied voltage by 90° (Chart 1). Since this series resonant circuit has a certain amount of A.C. resistance, however, and since the net inductance is quite low near resonance, the current is will lag the voltage e_8 by less than 90°. For explanation purposes, let us assume that series resonant circuit current is lags es by 45°. We therefore draw vector is lagging behind es by 45°, as indicated in Fig. 17A.

Regardless of the phase relation between i_8 and e_8 , the voltage drops across coil sections L_1 and L_2 will lead current i_8 by 90°, and since one voltage drop acts in the opposite direction from the other in so far as e_P is concerned, we can say that A.C. voltage e_1 leads circuit current i_8 by 90°, and A.C. voltage e_2 lags i_8 90°. We will leave Fig. 17A as it is, and redraw the vectors in Fig. 17B for the complete vector diagram. Vectors e_1 and e_2 are now placed on this diagram.

Again we add e_1 and e_P , taking phase into account, to find the net A.C. voltage acting upon diode section D_1 . We do this by drawing lines x and y in Fig. 17B to complete the parallelogram having e_1 and e_P for for two of its sides. The diagonal line drawn from the center of the vector diagram to the intersection of these dotted lines now represents the net A.C. voltage, $e_1 + e_P$, acting upon diode section D_1 .

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In the same manner we add A.C.

voltages e_2 and e_P in Fig. 17B, getting the somewhat longer vector $e_2 + e_P$ as the net A.C. voltage acting upon diode section D_2 . We can see immediately that vector $e_2 + e_P$ is longer than vector $e_1 + e_P$; this means that a higher voltage acts upon diode section D_2 than upon D_1 in Fig. 15, and a higher rectified current therefore flows through R_2 than through R_1 . The D.C. voltage drop across R_2 will therefore be greater



FIG. 17. Vector diagrams showing conditions in the discriminator circuit when the I.F. signal frequency is higher than the I.F. value of the receiver-

than the D.C. voltage drop across R_1 , and only a part of the drop across R_2 will be canceled out by the drop across R_1 . This leaves point δ negative with respect to ground and this negative D.C. voltage constitutes the D.C. control voltage applied to the oscillator control circuit. The higher the I.F. signal is above the I.F. value of the receiver, up to a certain limit, the higher will be this negative D.C. control voltage.

When the I.F. Signal Frequency is Low. If we repeat our analysis of the discriminator circuit in Fig. 15 and draw a vector diagram for the condition where the I.F. signal frequency is lower than the I.F. value of the receiver, we would find that net A.C. voltage $e_1 + e_P$ would be larger than $e_2 + e_P$, with the result that diode section D_1 produces a higher rectified voltage across R_1 than diode D_2 does across R_2 . The drop across R_2 would cancel out only a part of the drop across R_1 , with the result that the D.C. control voltage between point 5 and ground would have the same polarity as the drop across R_1 . The

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net D.C. control voltage is therefore positive with respect to the ground or chassis when the I.F. signal frequency is *lower* than the I.F. value of the receiver.

S Curve for Discriminator Circuit. The manner in which the D.C. control voltage produced by the discrimi-



FIG. 18. S-curve showing output characteristics of a discriminator circuit.

nator varies with the I.F. signal frequency is usually expressed by radio engineers in the form of a graph like that in Fig. 18. The curve on this graph is called an S curve. Note that a positive D.C. control voltage is produced for I.F. signal frequencies below the I.F. value of the receiver, and a negative D.C. control voltage is produced for I.F. signal frequencies above the I.F. value; this verifies the results of our analysis of the circuit in Fig. 15. Observe also that the D.C. control voltage increases quite uniformly as we go above or below resonance, up to points a and b. The part of the curve between these two points is the operating range for the discriminator; slightly beyond a and b, further deviations from the resonant frequency do not give additional increases in D.C. control voltage. As a result, there is insufficient control bevond these points and A.F.C. action is not complete. From a practical standpoint, this means that I.F. signal frequencies between points a and b will be properly corrected by the A.F.C. system, while frequencies outside this range will not be "pulled in" satisfactorily.

The Oscillator Control Circuit

The D.C. control voltage developed (sometimes the discriminator bv called the discriminator voltage) must be converted by the oscillator control circuit into an action which will incroase or-decrease the oscillator frequency the correct amount to make the I.F. signal frequency exactly equal to the I.F. value of the receiver. Since the frequency developed by the oscillator circuit is essentially controlled by the capacity of the oscillator tank circuit condenser and the inductance of the tank coil, this D.C. control voltage must be converted into a capacity or inductance which, when applied to the oscillator resonant circuit, will give the necessary change in frequency. In a practical oscillator control circuit, the vacuum tube is usually made to act as an inductance. for this produces a more uniform A.F.C. action over the entire band.

Before taking up an actual oscillator control circuit, let us see how we can tell when a vacuum tube is acting as a coil. We know that when a coil is connected to an A.C. voltage source, the current drawn by the coil will lag the voltage by 90°. Suppose we had a box with two terminals, and some unknown electrical device inside. If we connected this box to an A.C. voltage source and found that the current through the box lagged the voltage by 90°, we would immediately say that the box acted like a coil. As far as the A.C. voltage source is concerned, this box is behaving as an inductance.

When we increase the inductance of an ordinary coil, the reactance of the coil increases and as a result, the current through the coil is reduced. In other words, if the coil draws a low current, it has a high inductance; if the coil draws a high current, it has a low inductance. If, now, our imaginary box draws a high current which lags the voltage by 90°, we would say that this box has a low inductance; if the box draws a low current, we would say that it has a high inductance. You will soon see that we can consider the oscillator control circuit as an imaginary box having two terminals which are connected across one of the coils in the oscillator resonant circuit.

Now let us review a few fundamental facts about coils. When one coil is connected in parallel with the Note that the D.C. control voltage produced by the discriminator is applied to the control grid of the type 6J7 oscillator control tube through resistor R_2 . This D.C. control voltage, acting in series with the automatic C bias voltage produced by $C_{\rm C}$ and $R_{\rm C}$, determines the average D.C. plate current for the oscillator control tube. Turning now to the oscillator circuit, coil L_0 and tuning condenser section C_0 form the oscillator tuned circuit, with trimmer condenser $C_{\rm PD}$ serving





other, the inductance of the combination is less than that of either coil. Increasing the inductance of one of these coils increases the inductance of the combination, and likewise, decreasing the inductance of one coil decreases the inductance of the combination. An increase in inductance lowers the resonant frequency of a resonant circuit; a decrease in inductance increases the resonant frequency.

General Analysis of Oscillator Control Circuits. Now we are ready to investigate the behavior of the oscillator control circuit, to see how the D.C. control voltage produced by the discriminator can make this circuit act like an inductance and can change its effective inductance to correct for errors in tuning. A typical oscillator control circuit, along with the special superheterodyne oscillator circuit upon which it acts, is shown in Fig. 19A.

FIG. 19B. Vector diagram for oscillator control circuit.

as the oscillator padder. Condenser C_{g} is the R.F. grid coupling condenser, and serves together with R_{e} to produce the rectified D.C. grid voltage for the 6K7 oscillator tube. Coil $L_{\rm T}$ is connected between the plate and grid circuits of the oscillator, and therefore provides feed-back; the upper end of this coil is grounded for R.F. current through by-pass condenser $C_{\rm P}$. The oscillator tube receives its D.C. plate voltage through feed-back coil $L_{\rm T}$, while the oscillator control tube receives its D.C. plate voltage through oscillator tank coil L_0 .

Resistor R_1 and condenser C_1 in the oscillator control circuit are connected directly across the oscillator coil L_0 in so far as R.F. currents are concerned, and form what is known as the *phaseshifting network*. This network receives the full R.F. voltage developed by the oscillator circuit, for C_P is an R.F. by-pass condenser. The resistance of R_1 is so much greater than the reactance of condenser C_1 that we can consider this phase-shifting network as essentially a resistance. This means that the oscillator coil voltage e_0 will send through R_1 and C_1 an R.F. current i_R which is essentially in phase with e_0 .

Let us again resort to vector diagrams, to avoid the need for keeping phase relationships in mind while studying the circuit. We will use e_0 as our reference vector, and it is therefore drawn as shown in Fig. 19B. Vector $i_{\rm R}$ can now be placed, also along the reference line since it is in phase with e_0 .

The flow of R.F. current $i_{\rm R}$ through C_1 develops across this condenser an R.F. voltage e_{π} which lags $i_{\rm R}$ by 90° (see diagram L in Chart 1). We therefore draw vector e_{g} lagging behind $i_{\rm R}$ by 90°. This R.F. voltage $e_{\rm g}$ is applied to the control grid of the oscillator control tube through D.C. blocking condenser C_{κ} , causing A.C. plate current i_0 to flow through the oscillator control tube. This plate current will be in phase with the A.C. grid voltage, for this is a conventional amplifier circuit action in which a positive increase in the grid input voltage results in an increase in the plate current. We therefore draw vector $i_{\rm C}$ in phase with vector $e_{\rm g}$, as in Fig. 19B. Our vector diagram now shows clearly that R.F. current $i_{\rm C}$ lags oscillator coil voltage e_0 by 90°. If we consider that part of the circuit to the left of points 1 and 2 in Fig. 19A as a device having two terminals, we can readily see that the current $i_{\rm C}$ which is drawn by this device will lag by 90° the voltage e_0 which is applied to the device. We have thus shown that the oscillator control circuit acts as an inductance shunting the oscillator coil L_0 .

Now let us see how the D.C. control voltage produced by the discrimina-

tor will affect the value of this inductance shunting L_0 . If the D.C. control voltage which is applied to the grid of the oscillator control tube through R_2 is positive with respect to ground, the net bias on this grid will become less negative, increasing the D.C. plate current of the oscillator control tube. Since this 6J7 tube is operated on the curved portion of its E_{r} -I_n characteristic curve, the increase in D.C. plate current produced by a positive D.C. control voltage moves the operating point for the tube to a steeper portion of the E_{g} - I_{p} characteristic. This means that a given A.C. voltage eo on the grid of the tube will produce a larger A.C. plate current $i_{\rm C}$, making this stage act as a smaller inductance. If the D.C. control voltage is negative, the net C bias voltage on the 6J7 tube becomes more negative. This shifts the operating point nearer to plate current cutoff, and eo produces a low A.C. plate current, making this stage act as a larger inductance. Clearly, the D.C. control voltage produced by the discriminator serves to change the effective inductance of the oscillator control circuit.

Now, for the first time, we can consider the action of the entire A.F.C. system. When the receiver is properly tuned, so that the I.F. signal frequency corresponds to the resonant frequency of the discriminator resonant circuit, no D.C. control voltage is produced by the discriminator and the only C bias acting upon the oscillator control tube is that produced by $C_{\rm c}$ and $R_{\rm c}$. This bias sets the operating point for the tube at a point which allows a medium value of A.C. plate current i_0 to flow. The inductance corresponding to this A.C. plate current, acting in shunt with the oscillator inductance Lo, has been allowed for in the design of the oscillator, and consequently the A.F.C. system can

be considered inactive when the receiver is properly tuned.

When tuning is such that the I.F. signal frequency is below the I.F. value of the receiver, the discriminator produces a positive D.C. control voltage which increases the A.C. plate current and decreases the inductance effect of the oscillator control tube, thereby decreasing the shunt inductance across the oscillator tank coil and increasing the oscillator frequency.

When the I.F. signal frequency is higher than the correct value for the I.F. amplifier, the discriminator will produce a negative D.C. control voltage which makes the net negative C bias on the oscillator control tube more negative. The result is a decrease in A.C. plate current $i_{\rm C}$; this corresponds to an increase in the effective inductance shunting oscillator inductance $L_{\rm O}$, and this action of course lowers the oscillator frequency just enough to compensate for incorrect tuning.

The A.F.C. system cannot correct completely for errors in tuning, for then there would be no D.C. control voltage for correction purposes. The response of the I.F. amplier is always broad enough to allow for small errors in tuning, however, and consequently the A.F.C. action is satisfactory for all practical purposes.

Typical A.F.C. Circuits

Although the circuits in Figs. 15 and 19 have given the basic principles of A.F.C. systems, you will find a great many variations of these circuits. Before analyzing a number of typical circuits to familiarize you with these variations, let us consider briefly those components of each section which are to be found in all A.F.C. systems.

The discriminator section can be identified on the circuit diagram of a receiver having A.F.C. by the fact that it will have a double-diode vacuum tube, with the two diode plates connected into a tuned circuit having a split or center-tapped coil which is fed with the IF, amplifier output voltage both inductively and by a direct connection to the center tan. Each diode section thus gets two distinct I.F. voltages: 1, half of the I.F. output voltage of this resonant circuit; 2, the I.F. amplifier output voltage. The phase relationship between the voltages determines what the resultant I.F. voltage acting on each diode section will be.

In general, the oscillator control tube will have C bias voltages produced by two different sources: 1, by a cathode circuit resistor; 2, by the discriminator, feeding through a resistance-capacitance filter which keeps out R.F. components. Furthermore, the oscillator control circuit will have phase-shifting network, usually 8 made up of a resistor in series with a condenser, the combination being in shunt with the coil of the oscillator resonant circuit. The condenser in this phase-shifting network will have a very low capacity, ordinarily from about 2 mmfd. to 20 mmfd. Occasionally the grid-to-cathode inter-electrode capacity of the oscillator control tube will be used in place of a separate phase-shifting condenser.

With these general ideas in mind, we are ready to consider a few of the unique variations of the basic A.F.C. circuit just studied.

General Electric A.F.C. System. In Fig. 20 is shown the circuit diagram of the A.F.C. system and associated circuits used in General Electric Models E-101, E-105 and E-106 allwave superheterodyne receivers. The A.F.C. system follows very closely the fundamental circuit presented in Figs. 15 and 19, except that the discriminator in this General Electric circuit also serves as second detector and A.V.C. tube.

The output of the I.F. amplifier feeds into the discriminator circuit, through a conventional split-secondary I.F. transformer with which you are already familiar (Fig. 15). Notice, however, that the resistor between the cathodes of the doublediode tube is divided into three sections rather than two. Point 2 is the of low voltage rating, which provides an entirely satisfactory ground for I.F. and A.F. signals. The use of this condenser permits connecting point 1 through a 100,000-ohm resistor to a -3 volt terminal in the power pack voltage divider. In this way all A.V.C.-controlled tubes in the receiver get an initial C bias of -3volts, eliminating the need for automatic C bias resistors in some of the



FIG. 20. Circuit diagram of the A.F.C. system used in General Electric Models E101, E105 and E106 receivers.

midpoint of this resistor network. Between this point and the upper cathode (point 4) is a 470,000-ohm load resistor, while between point 2 and point 1, the lower cathode, is a 220,-000-ohm resistor in series with a 270,000-ohm unit, these together being approximately equal to the 470,-000-ohm resistor. The tap at point 3 serves to provide a lower A.V.C. voltage for one or more of the A.V.C.controlled stages in the receiver.

Ordinarily we would expect the circuit to be grounded to the chassis at point 1; instead, however, it is grounded through a 10 mfd. condenser cathode circuits. The 100,000-ohm resistor and 10 mfd. condenser also serve to keep power pack hum out of the discriminator circuit.

The cathode of the upper diode section is grounded through a 1 mfd. condenser, and the cathodes of both diode sections are therefore at ground R.F. potential. It is not essential that condensers of equal capacity be across each diode load resistor. The A.V.C. sources feed through 1.8-megohm A.V.C. filter resistors.

A resistance filter made up of two 100 mmfd. condensers and a 47,000ohm resistor is placed between point 2 and choke coil CH in order to place point 2 at ground potential for I.F. currents. These two condensers are not large enough to by-pass any audio currents to ground, but they do serve to keep I.F. currents out of the audio circuit.

Now locate the A.F.C. switch in the discriminator circuit. When this switch is closed, the A.F.C. voltage developed between point 4 and point 1 is shorted out, and no D.C. control voltage is applied to the oscillator control tube. The receiver now behaves like any ordinary superheterodyne receiver not having A.F.C. At first glance, one might believe that the A.F.C. switch also shorts out the A.F. and A.V.C. voltages; actually, however, the discriminator circuit becomes a full-wave detector, as shown in Fig. 21, when the switch is closed. With this connection, one diode will rectify one-half of each cycle of the I.F. signal, and the other diode will rectify the other half of each cycle. Both half-wave rectified currents are fed through the same parallel resistors, thus providing full-wave rectified voltages across these resistors.

The 1-mfd. condenser between point 4 and ground acts with the diode load resistors as a time delay which prevents immediate building up of the D.C. control voltage produced by the discriminator. The duration of this time delay is purposely made longer than the A.V.C. time delay, in order that the I.F. signal delivered to the input of the discriminator will reach its final value before A.F.C. circuit action begins. This is necessary because the D.C. control voltage produced by the discriminator is dependent upon the intensity of the I.F. signal as well as upon the amount by which it is off resonance (the greater the intensity of the I.F. signal, the greater will be the D.C. control voltage available for correcting mistakes in tuning, and consequently the A.F.C. system will be able to pull in strong stations even though the tuning dial setting may be considerably off from its correct value for that station).

The D.C. control voltage produced by the discriminator in Fig. 20 is fed to the control grid of the oscillator



FIG. 21. When the A.F.C. switch in the circuit of Fig. 20 is closed, the discriminator circuit is transformed into the conventional full-wave detector circuit shown here.

control tube through a 470,000-ohm circuit-isolating resistor. The parts between points a and b in the oscillator control circuit form the phaseshifting network. A .002-mfd. condenser is inserted between the customary resistor and phase-shifting condenser to prevent the D.C. plate supply source from acting upon the control grid of the oscillator control This extra condenser tube. has negligible reactance at R.F. signal frequencies, and therefore serves merely as a D.C. blocking condenser.

The oscillator control circuit to the left of points a and b acts as an inductance whose value depends upon the polarity and magnitude of the D.C. control voltage provided by the discriminator; thus effective inductance is shunted across the coil in the oscillator resonant circuit, the circuit being throug) a 30 mfd. condenser which serves as a D.C. blocking condenser, and through an 8,200-ohm resistor which has the effect of equalizing over the entire tuning range of the receiver the amount of shunt inductance applied to the oscillator resonant circuit.

With these facts in mind, you should have no difficulty in completing an analysis of this General Electric A.F.C. system. It reacts in exactly the same way as the circuits in Figs. 15 and 19 to I.F. signals which are above or below the I.F. value of the receiver.



FIG. 22. Types of S-curves obtained with discriminstor transformers having various degrees of coupling.

Discriminator Transformer Design Problems. The conventional discriminator transformer used in the circuits of Figs. 15 and 20 require careful de-Weak coupling between the sign. primary winding and the split secondary winding makes these two tuned circuits highly selective, giving the Scurve marked A in Fig. 22. The steepness of the linear region of this curve indicates that stations will be pulled in very close to the I.F. value (to point 3); the small frequency difference between the ends of the linear region means that manual tuning must be quite accurate before A.F.C. will take hold, and the relatively low magnitudes of D.C. control voltages provided by curve A may not be sufficient for full correction.

Approximately critical coupling gives S-curve B, indicating greater D.C. control voltages and a greater frequency range over which A.F.C. will pull in a station; in fact, the linear range may be too wide, allowing an undesired station to be dragged over an adjacent desired station. The linear section of the curve is less steep, indicating that A.F.C. will not correct as completely as with curve A, but a deviation of a few hundred cycles from resonance is ordinarily quite permissible.

When coupling is greater than the critical value, the two tuned circuits interact, giving a double-peak response curve which results in the undesirable S-curve at C in Fig. 22. This curve tells us that I.F. signals which are outside of the region between points 1 and 2 will be pulled in to a frequency corresponding to these points rather than to the correct I.F. value at point 3.

The curves in Fig. 22 show clearly that the design of a discriminator transformer must be a compromise between the various desired performance characteristics. In addition, the receiver must have a good A.V.C. system, in order that signals which are received at different signal levels will produce almost identical voltages in the discriminator circuit. If this were not true the amount of correction provided for mistuning would vary with the strength of the signal picked up.

Silvertone (Sears Roebuck) A.F.C. System. The manufacturers of Silvertone receivers, as well as a number of other manufacturers, have overcome these discriminator transformer design problems by using a tripletuned circuit ahead of the discriminator, as shown in Fig. 23. The three resonant circuits, L_1 - C_1 , L_2 - L_M - C_2 , and L_3 - L_4 - C_3 are all peak-tuned to the I.F. value of the receiver. Circuit L_3 - L_4 - C_3 is designed with the correct Q factor to give the desired S characteristic, and the coupling between L_{M} and L_{3} - L_{4} is made close enough to give the required voltages across L_3 and L_4 for the diode sections.

Another unique feature of this Silvertone A.F.C. system is the absence of the choke coil ordinarily found between points 1 and 2. In this particular circuit the choke coil is unnecessary because the required voltage is developed by the middle resonant circuit $(L_2-C_2-L_M)$ which is dilator control circuit. The connection is through a 1-megohm resistor and .1 mfd. condenser which serve as the A.F.C. filter. The time constant of this filter is greater than that of the A.V.C. filter, as is required.

The oscillator control circuit in Fig. 23 is of conventional design, with R_N



FIG. 23. Circuit diagram of the A.F.C. system used in the Silvertone model 4587 receiver.

rectly connected between points 1 and 2. Point 2, being negative with respect to ground and having a potential which is proportional to the rectified current flowing through the lower diode load resistor, serves as the A.V.C. terminal. The 1-megohm resistor and .05 mfd. condenser make up the A.V.C. filter which removes A.F. and I.F. components from the A.V.C. voltage and at the same time determines the time constant of the A.V.C. system. Point 2 also serves as the A.F. voltage supply terminal, with the .05 megohm resistor and the two .00025 mfd. condensers keeping I.F. signals out of the audio amplifier. Terminal 3 on the diode load resistor network is, of course, the point from which the D.C. control voltage for A.F.C. purposes is fed into the osciland C_N serving as the phase-shifting network. The receiver oscillator is of the tuned plate type, with the phaseshifting network shunting the oscillator tuning coil.

A somewhat similar discrimination circuit is also used in one RCA A.F.C. system; the essential difference between the RCA and Silvertone circuits is shown in Fig 24. The first two tuned circuits in the discriminator transformer are the same in both cases, the difference being in the third tuned circuit. Note that in Fig. 24 coil $L_{\rm M}$ induces a voltage in the splitcoil arrangement L_3 - L_4 , rather than in two separate coils as was the case in Fig. 23. Across L_3-L_4 is fixed condenser C_3 and variable inductance L, having an iron dust core whose position can be varied in order to change

trodes from stray fields (stray radiated signals) of other parts. Without a shield, the set would probably squeal at each station as you tuned across the dial.

Tube Bases. The tubes in this receiver all have octal bases, which means that each one can have as many as eight prongs. Some of the tubes require all eight prongs for connecting purposes, either because the tubes have extra grids or because two or more sets of electrodes are combined in one tube.

When a tube requires less than eight connections to its socket, some of the prongs are omitted during manufacture without changing the positions of the others. Thus, the 6X5



FIG. 7. Removing the shield from a tube.

rectifier tube taken from our set and shown in Fig. 8 has only six prongs.

Tube Numbers. Each of the tubes in our receiver has an identifying number of its own, printed right on the glass envelope or on the base of the tube. Furthermore, each tube is intended for a particular socket on the chassis base of this receiver. Since an octal-base tube will fit in any octalbase socket, the sockets must be identified in some way so as to make sure



FIG. 8. This is what you see when you remove a 6X5 tube from a receiver and look at its base.

that the correct tube will be placed in each socket.

Socket Markings. The number of the tube intended for each socket has been clearly printed on the chassis base of our receiver, in the manner shown in Fig. 9. This illustration shows how the set would appear from the rear if only the tubes were above the chassis. The functions of the tubes are indicated by notes on the illustration.

Sometimes tube sockets themselves are marked. More often, however, there are no markings anywhere near the sockets. Instead, there is a diagram attached to the chassis or the inside of the cabinet, showing which tube goes in each socket.

In a few sets, there will be no markings at all. You can either mark all the sockets yourself before taking out the tubes, or take out only one tube at a time for testing.

Removing Tubes. To remove a tube for testing or replacement, pull it straight up while wiggling it slightly

the inductance of L and tune the discriminator circuit. Coil section L_3 also has an adjustable iron dust core which can be adjusted during alignment of the receiver to make the inductance of L_3 exactly equal to the inductance of L_4 and thus make both _ coils develop the same I.F. voltage for the diode section.

Effect of Preselector Upon A.F.C. Action. Up to this point we have neglected the effects of the preselector upon the action of the A.F.C. sys-Actually, however, the selectem. tivity of the preselector is quite important with relation to that characteristic of an A.F.C. system which is known as "dragging." Consider, for example, a receiver having an S curve which indicates the ability to correct for signals as much as 5 kc. off tune. When a station is properly tuned in, then gradually tuned off resonance. the A.F.C. circuit will correct the oscillator frequency satisfactorily for about 5 kc. off resonance. Tuning farther off than this does not remove the discriminator control voltage entirely, as you can see by referring to the S curve in Fig. 18; actually the D.C. control voltage may be produced for a considerably greater deviation from resonance, but it will be insufficient to correct completely for the offtune condition. This action of an A.F.C. system in making a partial correction outside of the efficient operating range of the system is referred to by engineers as "dragging."

When two stations are close together in frequency, and one of the stations is tuned in, it is perfectly possible that a receiver having A.F.C. and a broad preselector might not be able to pick up the other station. The station originally tuned in is held by the A.F.C. system because of excessive dragging action. With this broad preselector, the only way to hear the adjacent station would be to cut off

temporarily the A.F.C. action of the receiver in order to allow the adjacent station signal to take hold of the A.F.C. system; a highly selective preselector, however, *would not allow* an A.F.C. system to hang onto a powerful local station when the receiver is tuned to a weaker adjacent station.

Here is another interesting result of dragging action; when a receiver having A.F.C. is tuned carelessly to a desired weak station, so that the desired signal is being dragged in from a position considerably off tune, any fading of this desired signal may allow an adjacent station to take hold.

The remedy for excessive dragging involves improving the selectivity of the preselector enough so that when the tuning condensers are outside the desired range of A.F.C. action, the reduction in R.F. gain will reduce the I.F. signal enough to make the discriminator release its control over the oscillator control tube. Strictly speaking, the preselector should be bandpassed so it will tune broadly in the frequency range over which A.F.C. is to act, but will have a sudden reduction in gain (high selectivity) immediately outside of this band-pass region.

Band-pass preselectors are rarely used in receivers, but sufficient selectivity should be incorporated in the preselector to prevent annoying dragging action. Because it is not easy to get high preselector selectivity in the high frequency bands of an all-wave receiver, A.F.C. action is often omitted in these bands. Any A.F.C. circuit is most effective for strong stations in the broadcast band.

Philco Push-Pull A.F.C. System. The automatic frequency control system used in some Philco receivers is unique in that when the receiver is properly tuned or the A.F.C. system is shorted out, the receiver oscillator circuit is entirely independent of the A.F.C. system. It is therefore possible to align the oscillator and preselector independently of the A.F.C. system, whereas in the systems previously described the shunt inductance of the oscillator control tube always had to be considered when aligning the oscillator trimmer condensers.

A typical example of this unique Philco push-pull A.F.C. system, that used in the Philco model 37-9 allwave superheterodyne receiver, is shown in Fig. 25; it is also known as a magnetic tuning system. As you can see, the type 6K7G tube in the final I.F. amplifier stage feeds into two entirely separate circuits, a 6Q7G second detector-A.V.C.-first A.F. tube circuit (not shown) and the conventional 6H6G double-diode discriminator circuit.

The two diode sections of the discriminator tube each receive two I.F. voltages in the usual manner: the I.F. voltage existing across the tuned plate circuit of the 6K7G tube is applied between the center tap of the split secondary winding and the cathodes of the 6H6G tube, and the I.F. currents flowing in the split secondary (due to the voltage induced by current in the tuned plate circuit) produce across the coil sections the other I.F. voltage acting on each diode section. The resulting rectified current flow develops the required D.C. discriminator control voltage across the two 2-megohm resistors connected between the cath-The polarity of this voltage odes. depends, of course, upon whether the I.F. signal frequency is above or below the correct value, and the magnitude of this D.C. voltage depends upon the amount of error in tuning. This D.C. control voltage is applied across two 1-megohm resistors having their midpoint grounded, and hence points x and y will always be equal to each other in potential and of opposite polarity with respect to ground. In this manner the discriminator output voltage is divided into two equal voltages of opposite polarity.

During off-tune conditions these equal and opposite voltages at x and y are fed to points a and b, at the input of the oscillator control tube. Note that there is a 490,000-ohm re-



FIG. 24. Discriminator transformer connections used in RCA model 812K receiver.

sistor in each lead, with a .3 mfd. condenser shunted across the leads: these filters serve to remove all I.F. and A.F. signal components and at the same time delay the discriminator action enough to make it slower than A.V.C. action. The two .15 mfd. conwith common terminals densers grounded serve to maintain a balance between the filtered A.C. currents. Switch SW_1 is the A.F.C. ON-OFF switch; when in the OFF position it grounds points a and b. Switch SW_2 closes only when the dial tuning mechanism is operated, and shorts points a and b to prevent A.F.C. from dragging one station beyond the dial setting for another station.

Observe now that a type 6A8G pentagrid converter tube serves as the first detector-mixer-oscillator tube for the receiver, with the first grid serving as the oscillator control grid and the second grid acting as the oscillator plate. The oscillator tuned circuit, made up of L_1 , C_1 , C_3 and the 85ohm resistor, is connected between the first grid and the chassis. The necessary feed-back is obtained by connecting the oscillator plate electrode through a 250-mmfd. condenser to the lower end of coil L_1 ; current flow through this section of L_1 induces the feed-back voltage in the main section of L_1 . C_3 serves as the oscillator low frequency padder and C_2 as the high frequency trimmer. Rectified grid current flowing through L_1 and a-32,000-ohm resistor in the oscillator tuned circuit produces across this reinto the discriminator circuit. This R.F. voltage $e_{\rm R}$ produces R.F. plate currents i_2 and i_3 which are *in phase* with $e_{\rm R}$, are equal in value when both tube sections have the same C bias voltage, and flow in opposite directions through the two halves (L_2 and L_3) of the secondary winding of the oscillator control transformer.



FIG. 25. Circuit diagram of the push-pull A.F.C. system (also known as magnetic tuning) used for the broadcast band on the Philco model 37-9 all-wave receiver. There is no A.F.C. action on short-wave bands in this set.

sistor a self-adjusting negative C bias which acts in series with the fixed C bias produced by the oscillator cathode resistor.

Locate the oscillator tuned circuit, shown in heavy lines in Fig. 25 and made up essentially of L_1 , C_1 , C_3 and an 85-ohm resistor. The same R.F. current i_1 flows through all of these parts, developing across the 85-ohm resistor an R.F. voltage $e_{\rm R}$ which is in phase with the current.

R.F. voltage $e_{\rm B}$ is applied to the two control grids of the 6N7G oscillator control tube through 110 mmfd. condensers. Note that 99,000-ohm resistors connected to these grid leads prevent the R.F. voltage from feeding

From your study of coils you will recall that increasing the amount of flux through a coil increases its inductance, and decreasing the flux lowers the inductance. If current i_2 flowing through L_2 produces a flux which increases (aids) the flux produced by i_1 through L_1 , then i_8 flowing in the opposite direction through L_3 will produce a flux which decreases (opposes) the flux produced by i_1 through L_1 . (This holds true only when all three flux-producing currents are in phase. as they are in this circuit.) Current i_2 thus has the effect of increasing the inductance of L_1 in the oscillator tuned circuit, while i_3 decreases the inductance of L_1 .

When i_2 and i_3 are equal, their effects upon the inductance of L_1 will be equal and will cancel each other. This proves clearly that when the A.F.C. system is inoperative, or when the receiver is properly tuned, the oscillator control tube has no effect whatsoever upon the oscillator tuned circuit. moving the operating point to a less steep region and reducing the A.C. plate current for that section. The D.C. control voltage thus makes i_2 different from i_3 . The effects of the larger current upon the inductance of L_1 are not altogether cancelled out by the effects of the smaller current; the result is a net change in the induc-



CHART 1. This chart gives you important facts about the phase relationships of voltages and currents for resistors, coils, condensers, and transformers in A.C. circuits.

As you already know, improper tuning of a receiver causes D.C. voltages of equal value but opposite polarity with respect to ground to be developed by the discriminator; these are applied to the grids of the 6N7G oscillator control tube through 99,000ohm resistors, making one grid less negative and the other more negative than the bias value determined by the 700-ohm cathode resistor. Making the net C bias of one triode section less negative moves the operating point on the E_{g} - I_{p} characteristic to a steeper region, increasing the A.C.plate current for that section; at the same time the net C bias on the other triode section becomes more negative, tance of L_1 , and this changes the frequency of the oscillator. In other words, one current *pushes* up and the other *pulls* down the inductance of L_1 . The manner in which leads a and b are connected to the control grids of the 6N7G tube determines whether this control will be in the proper direction; if the control is such as to exaggerate errors in tuning, reversing the leads will correct the trouble.

Adjusting A.F.C. Systems

Fortunately for Radiotricians, the adjustment of an A.F.C. system is quite simple. The only instruments needed are an ordinary signal generator capable of producing the I.F. value of the receiver, and a sensitive low-range (0-10 volts) voltmeter. At least a 5,000 ohms-per-volt meter should be used; if this is not available, use an ordinary vacuum tube voltmeter or one which you assemble yourself from a type 31 tube and batteries, using a milliammeter as the indicator.

The signal generator is set at the I.F. value of the receiver and is connected to the input of the I.F. amplifier. The sensitive voltmeter is connected across the D.C. output terminals of the discriminator (points 1 and 4 in Fig. 20; between the chassis and point 3 in Fig. 23; between points a and b in Fig. 25). The I.F. amplifier stages are now aligned in the usual manner for a maximum reading of the voltmeter. (Do not touch the trimmer across the split secondary of the discriminator stage at this time unless the voltmeter deflection is too small for an accurate adjustment; in this case, adjust the trimmer in either direction just enough to get a suitable deflection.) When adjusting the tuned circuit ahead of this split secondary winding, it is a good plan to move the signal generator to the input of the previous tube; adjust this discriminator input circuit carefully for maximum output.

Now, without changing the voltmeter or signal generator connections, adjust the trimmer condenser in the split coil circuit for *sero meter reading*. This adjustment is quite critical, so be sure that the minimum reading is secured when the adjusting tool is removed. This is all there is to the adjustment of a conventional A.F.C. circuit.

You might think that with so simple an adjusting procedure, the information given in the first part of this lesson is unnecessary for the practical man. This is decidedly not true, however, for an understanding of how A.F.C. systems work is quite necessary when trouble develops due to failure of various parts in the system. If your work will involve the servicing of receivers having automatic frequency control, you will at some time or other find use for every bit of the information given in this lesson.

2. To correct the inaccuracius in this mechanical action of the Tuning mechanism and to automatically compensate for prequency drift.

Lesson Questions

Be sure to number your Answer Sheet 31FR-1.

Place your Student Number on every Answer Sheet.

Never hold up one set of lesson answers until you have another ready to send in. Send each lesson in by itself before you start on the next lesson. In this way we will be able to work together much more closely, you'll get more out of your Course, and you will receive the best possible lesson service.

- 1. Where are meter-type tuning indicators ordinarily connected in a receiver having A.V.C.? To the plate supply lead going to the AUC.
- 2. Why is an A.F.C. system essential in a receiver having an electro-mechanical (motor-driven) automatic tuning system?
- 3. What voltage in a receiver is used to control a cathode ray tuning indicator? The DC willinge durily and any tuning indi-
- 4. How does core saturation affect the reactance of the secondary winding of the General Electric Colorama tuning reactor?
- 5. Name the two essential sections of an A.F.C. system.
- 6. In a receiver having A.F.C., what voltage serves to change the effective inductance of the oscillator control circuit?
- 7. How can you identify the discriminator section on the circuit diagram of a receiver having A.F.C.?
- 8. Will a highly selective preselector allow an A.F.C. system to hang on to a powerful local station when the receiver is tuned to a weaker adjacent station?
- 9. When the Philco push-pull A.F.C. system is inoperative or when the receiver is properly tuned, what effect does the oscillator control tube have upon the oscillator tuned circuit?
- 10. When adjusting an A.F.C. system a signal generator set at the I.F. value of the receiver is fed into the I.F. amplifier and a sensitive D.C. voltmeter is connected across the D.C. output terminal of the discriminator. How is the trimmer condenser in the split-coil discriminator circuit adjusted as regards the meter reading?

COURTESY

Truly big men are always courteous. It is only "small" men, men with inferiority complexes, who are rude or thoughtless. And smaller than small are those who are over-courteous to their superiors and intentionally rude to those over whom they have some authority.

Practice courtesy in all your contacts, business as well as social. Establish courtesy as one of your life habits. Be polite and kind to every one you meet, rich and poor alike, and soon courtesy will become second nature for you.

The best place to test yourself is right at home. Are you always courteous to members of your immediate family, or do you shout at them on the slightest provocation? Are you considerate of their feelings, or do you delight in saying things and doing things which you know will hurt them?

If you develop the habit of being courtcous to your own folks, your away-from-home courtesy will ring true. It won't appear "put-on," as is so often the case when a man reserves his courtesy for only special occasions.

J. E. SMITH





Study Schedule No. 32

For each study step, read the assigned pages first at your usual speed, then reread slowly one or more times. Finish with one quick reading to fix the important facts firmly in your mind, then answer the Lesson Questions specified for that step. Study each other step in this same way.

1. What a Circuit Diagram Tells; Principles of Radio Servicing Pages 1-7

Circuits need not be completely analyzed; two groups of receiver defects; Surface Defects; Dead Receiver; Improperly Operating Receiver; troublelocating questions; Distortion at High Volume Levels; Hum; Alignment Procedures; Subjects for Further Study; example for study is Philco Model 38-7 receiver. Answer Lesson Questions 1, 2 and 3.

2. Identifying Stages on a Schematic Diagram; Analyzing

Each Stage on a Schematic Diagram - - - - Pages 7-12 Read diagrams from left to right; identifying clues on diagrams; stageidentifying examples: Oscillator-Mixer-First Detector; I.F. Amplifier; Second Detector; A.V.C. Tube; First A.F. Amplifier Stage; Output Stage; Power Pack. Answer Lesson Questions 4 and 5.

3. Identifying Stages on the Chassis - - - - - Pages 12-20 Use of tube layout diagrams; identifying the circuit; locating the rectifier tube and a.f. output tube; identifying other tubes; General Suggestions. Answer Lesson Question 6.

4. Locating Parts with a Pictorial Diagram; Locating Parts Without a Pictorial Layout Diagram; Tracing Tube

Electrode Circuits - - - - - - - Pages 20-24 Locating electrolytic filter condensers and identifying their leads; locating a resistor; use of circuit diagram alone for locating parts and checking continuity; Plate Circuits; Screen Grid Circuits; Control Grid Circuits; Suppressor Grid Circuits; Diode Detector Circuits. The important thing in this step is to refer to the diagrams as you study, so as to get practice in locating parts quickly with the aid of diagrams. Answer Lesson Question 7.

5. General Rules for Checking Continuity of Electrode Circuits

in A.C. Receivers; How to Locate Trimmer Condensers;

Appraising Receiver Performance; Schematic and

Chassis Wiring - - - - - - Pages 24-29 Rule for positive electrodes; rule for negative electrodes; rule for diodes; identification of trimmers; General Trimmer-Locating Suggestions; how performance is affected by the number of tuned stages, number of tubes, type of circuit, use of r.f. stage, type of loudspeaker and presence of band-pass circuits; practical chassis wiring. Answer Lesson Questions 8, 9 and 10

☐ 6. Mail your Answers for this Lesson to N.R.I. for Grading.

7. Start Studying the Next Lesson.

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How To Use Circuit Diagrams and Locate Chassis Parts

What a Circuit Diagram Tells

AS YOU know, a schematic circuit diagram is a symbolic means for showing the electrical connections in a radio set. A schematic diagram tells very little about the construction of each part, how the parts should be adjusted, where the parts are located on the actual chassis, or how the connections are actually made between various points; nevertheless, to a man possessing the fundamental training in radio theory and practice which you are now acquiring, this diagram can yield information of great value in connection with the servicing and maintenance of radio equipment.

There is no more need for a Radiotrician to analyze a complete receiver circuit diagram when repairing a defect, than there is need for a carpenter to study the complete building plans for a new home when repairing the front porch railing. Once you become familiar with the general principles of radio servicing, you will agree that a circuit diagram is a "reference text" rather than a "study text."

Principles of Radio Servicing

When you tackle a defective receiver, you know that it must belong to one of the two general trouble groups in which all defective radio receivers can be placed: 1, the receiver is dead—it does not play at all; 2, the receiver plays improperly—it may howl, squeal, distort, lack selectivity, lack sensitivity, be noisy, have a faulty automatic tuning or A.F.C. system, or have any one of a host of other defects which are considered elsewhere in the Course.

Surface Defects. Occasionally the

trouble which results in a dead or improperly operating receiver is not in the chassis at all, but on the outside or on the *surface*, where it can be readily seen and corrected. Obviously you will want to check for surface defects before removing the chassis.

The practical radio serviceman generally starts in by checking the antenna system and the power line



Making a continuity test between the top caps of two tubes on the Philco model 38-7 receiver, using the multimeter section of a combination signal generator and multimeter.

cord, making sure that all tube caps are in position and not touching the shields, making sure that all extension wires and cable plugs are in place, and noting whether any tubes are obviously defective. He may even remove each tube and check it in a tube tester, to see if a simple replacement of a defective tube will clear up the trouble.

When a check for surface defects in this manner proves unsuccessful, one trouble-locating procedure is followed for the case of a dead receiver, and an entirely different pro-

orld Padio Histo



FIG. 9. This is how the back of our superheterodyne receiver chassis would look if only the tubes were visible.

from side to side. Tube sockets are purposely made to grip tube prongs tightly, so as to provide good electrical contact and prevent tubes from bouncing out during shipment of a receiver. Thus, it may take quite a bit of force to pull out a tube.

Here is a practical tip which may prove mighty useful when you are trying to remove a tube from a crowded corner of a radio chassis. While pulling and wiggling the tube with your right hand, insert a screwdriver under the tube base with your left hand and twist the screwdriver so as to pry up the tube. Of course, the receiver must be turned off when this is done, for otherwise the screwdriver might short some power supply circuit and cause damage.

Tubes Get Hot. All radio tubes heat up when in operation, but some types of tubes get much hotter than others.

The tubes used in battery and portable receivers stay the coolest. Allmetal tubes, on the other hand, sometimes get so hot in normal use that you can actually burn your fingers on them.

The tubes in our superheterodyne receiver all use glass envelopes, which in some cases get almost as hot as metal envelopes. Allow a minute or so for the tubes to cool before attempting to remove them, or use a handkerchief or piece of cloth to protect your hand when removing hot tubes.

This precaution applies particularly to the audio output tube and the rectifier tube in a receiver, for these two tubes usually get the hottest. The reason is that the audio output tube has a higher plate current than any other tube except the rectifier tube. As a matter of fact, the rectifier tube handles all the plate and grid currents drawn by all the other tubes in the receiver.

Testing Tubes. When a receiver comes into a radio service shop for repair, tubes are usually the first things which the serviceman checks. He removes the tubes from the chassis and tests them, one at a time, in a special tube-testing instrument like that shown in Fig. 10. cedure is chosen for an improperly operating receiver. In both of these procedures the schematic circuit diagram is brought into use.

Dead Receiver. When anything goes wrong, the natural first step is to ask yourself what the trouble could This applies to any machine, to be. any human being, and to any action, as well as to the radio receivers under discussion. We therefore ask: "Why does a receiver become dead?" There are a number of answers-one or more tubes may be defective, some part in the power pack or in the voltage supply system for the receiver may have opened up or become shorted, a signal circuit part may have opened or shorted, or a connection may have opened somewhere.

With a dead receiver, then, our first task is the locating of the defective tube, part or connection. Do we have to check all parts and connections until we find the one which is defective? Not if we reason out this problem a bit first. You know that a carrier signal modulated with an intelligence signal enters the antenna circuit of a receiver, with this intelligence signal working its way through a chain of stages to the loudspeaker. The receiver is dead because one or more links in this chain have failed. Radio men say that one or more stages have become defective; once they locate this defective stage, they have limited the task of locating the defective part to a smaller portion of the receiver.

As you will learn elsewhere in your Course, there are two ways of locating the defective stage in a dead receiver; we will present them here only briefly. In the *dynamic stage-bystage elimination test*, a modulated signal generator is connected to the input of the second detector; if the modulation signal is heard in the loudspeaker, you know immediately that the detector and the audio amplifier are free of trouble. If no signal is heard, headphones are connected in turn to the output of the detector and to the output of each following audio stage. Failure to hear the output signal in the headphones at any stage identifies that stage as being defective, assuming that the power supply has been checked and found to be performing properly.

If the detector and the stages following it are found to be operating properly, the signal generator is advanced toward the antenna one stage at a time; the signal should become louder as this is done. Failure to hear the signal identifies the defective stage.

<u>The simplest and speediest test for</u> isolating the defective stage in a dead receiver is the circuit disturbance test. This depends upon the fact that when a tube is pulled out of its socket and quickly returned, or when the grid of a tube is touched or shorted to the chassis, or when the grid cap connection to a tube is removed and returned, a sharp change in the plate current for the tube takes place. This sudden change or shock is relaved through the receiver and is heard in the loudspeaker as a click, a thud or a squeal provided that all stages in the path of this disturbance are operating properly. Starting at the last audio stage and working toward the antenna stage, one of these methods for producing a disturbance is employed on each tube in turn. When you reach a tube which produces no indication in the loudspeaker while being shocked, you have isolated the trouble to that stage.

Once the defective stage is isolated, it is given a thorough check either with an ohmmeter or a voltmeter in a manner explained elsewhere in the Course, after making sure that the tube is good. Supply voltages for the tube are usually checked first.

Obviously you must know something about the circuit of a receiver in order to be able to isolate the defective stage and then isolate the defective part in that stage. You must be able to identify the second detector tube, which is the usual starting point for a stage isolation procedure. You must be able to locate the first, second and power audio tubes so you can follow the chain of stages in the proper order from the second detector to the loudspeaker; likewise you must be able to locate the I.F. and R.F. amplifier tubes in their proper order from the second detector to the antenna. You should know where toconnect a voltmeter in order to check the main power supply. Once the defective stage is located, you must be able to locate and identify the various parts in the grid, plate, screen grid and suppressor grid circuits of the tube. You may be able to determine all these things by a study of the chassis itself, but this will take time; a schematic circuit diagram gives this information far more readily. If the service manual of the receiver also contains a tube layout diagram, showing the positions of the various tubes on the chassis and identifying their functions, your task is, further simplified.

Receiver. Improperly Operating Expert Radiotricians will tell you that a dead receiver is generally easier to service than an improperly operating receiver. The true value of theoretical training, modern technique and practical experience becomes evident when it is necessary to service a receiver which squeals (oscillates), blocks, distorts, hums, is noisy, cuts off intermittently, lacks selectivity, lacks sensitivity, or lacks output volume. The value of circuit diagrams becomes increasingly more evident as servicing problems become more complicated.

A customer may bring in a receiver for repair, complaining that he can't understand the announcements or speeches as reproduced by the loudspeaker. After making sure that surface defects are not the cause of this distortion, a Radiotrician will automatically ask himself these questions:

1. Under what conditions of receiver operation does distortion ordinarily occur? Does the receiver distort only when the volume level is high, only when the volume level is low, or at all volume levels and all settings of the receiver controls?

2. What are the general reasons for distortion which occurs under the conditions noted?

3. What are possible causes, in this particular receiver, for the type of distortion observed? (Reference text "Radio Receiver Troubles—Their Cause and Remedy," which was sent you some time ago, covers many of the common causes for various types of distortion. This reference text will help you develop an ability to reason from observed effects like distortion to the logical causes.)

When a serviceman cannot answer these questions, he is compelled to test all stages and parts, using some systematic method of defect isolation. ~ The checking of operating voltages for all tubes is an important part of this procedure; the correct voltages are usually given in the service manual.

Distortion at High Volume Levels. Let us assume that the receiver in question distorts only when the volume level is high. If it is an inexpensive midget table model receiver, this distortion may be natural and unavoidable, especially if it is most serious when powerful local stations are tuned in. Experience is a great help in determining whether or not an actual defect exists. Reference to the circuit diagram of the receiver will

3

often tell whether or not the receiver is capable of handling high volume levels. If you decide that the receiver is operating as well as can be expected for that particular type of circuit, you will have to explain the situation to the customer and point out that the volume level should be kept below the point at which distortion begins.

If the receiver is of normal design and you decide that it should handle the maximum volume level without distortion, the next question to answer is: What can cause distortion when the volume level is high? The Radiotrician knows that possible causes for this effect are overloading of the detector, overloading of the audio output tube, overloading of the loudspeaker, weak tubes (low emission), low supply voltages, or a defective loudspeaker. With these probable causes in mind, you should now proceed to eliminate them one by one until you reach the true cause of the trouble. Test the tubes first of all, then' check the main supply voltage and the individual supply voltages for each suspected tube. You may have to refer to a circuit diagram in order to determine where the voltmeter connections should be made for each test.

Plate and screen grid supply voltages are usually measured between these electrodes and the cathode of the tube, while a <u>C</u> bias voltage is measured between the cathode and the chassis, as a rule. Quite often, manufacturers will specify, in service manuals, all electrode voltages with respect to the chassis. Reference to the schematic circuit diagram will tell you exactly where connections can best be made to measure a particular voltage.

Can you quickly locate the plate, screen grid and cathode terminals on an actual tube? Few people can, for there are hundreds of different types of tubes in use today, each with different connections between the electrodes and the terminal prongs. Receiver manufacturers sometimes show tube sockets in pictorial form in the service manual, either on the schematic circuit diagram itself or on a sepfarate chassis diagram. Tube manufacturers likewise supply socket connection diagrams for all tubes, so there should be no trouble in securing a tube diagram to serve as a guide.

When there are two or more stages in a receiver employing the same type of tube, how can you locate one of these tubes on the circuit diagram? The answer is quite simple; examine a few of the parts in the chassis which are connected to the tube in question, then determine which tube in the schematic diagram has these same parts connected to it.

Now you can check the values of each part in the defective stage against the values specified on the circuit diagram; check for changes in the resistance of a part and for shorted or leaky condensers, as any of these defects will alter the supply voltage enough to cause distortion when the volume level is high.

If you have followed the servicing procedure up to this point without locating the defect, refer to the schematic circuit diagram and determine whether the receiver has A. V.C. If the volume control is connected into the input of the audio system, you can be pretty sure that some of the preceding I.F. and R.F. amplifier stages are A.V.C controlled. Assuming this to be the case, you can be fairly certain that R.F., I.F., and second detector signal and supply circuits are all in order since there is no distortion at low volume levels on the particular receiver being analyzed.

You think a bit-could there be

overloading of an audio amplifier stage? As soon as you think of a reasonable cause of trouble, check up on it. An easy way to locate the overloaded stage is to turn up the volume control while tuned to a strong local station, so that distortion is clearly evident, and then connect headphones across the output of each audio stagein turn, starting with the second detector and ending at the loudspeaker terminals. Reference to the schematic circuit diagram and perhaps to the pictorial layout diagram will tell you where to make the headphone con-The stage at which disnections. tortion is first noticed in the headphones will very likely be the overloaded stage. If no distortion is first noticed in the headphones up to the loudspeaker, then you can be pretty certain that there is a defect in the loudspeaker or its input circuit.

Hum. Now let us consider a receiver which has excessive hum. Naturally you turn on the set and make the usual inspection for surface defects. Let us assume that while doing this you note one tube with a blue glow between the cathode and plate. By noting the number of this tube, by referring to the parts layout diagram in the service manual, or by noting that there are no grid electrodes between the cathode and plate in the tube, you identify it as a rectifier. Obviously you have located a surface defect, for these tubes (with the exception of mercury vapor rectifier tubes) should not glow when operating properly; there is no sense in looking farther for other causes of hum until this trouble is cleared up,

A blue glow inside a vacuum type rectifier tube indicates the presence of gas; we say that the tube is gassy. The question is: Why did the tube become gassy? A logical answer is normal aging of the tube with constant use, for rectifier tubes are often

operated very nearly at their maximum capacity in receivers. On the other hand, there may be a short in the power pack which is making this rectifier tube deliver higher than normal current. It would not be safe to insert a new tube until the power pack has been examined for possible defects. Again you would refer to the circuit diagram, noting what parts are in the power pack circuit and noting what their values should be. From your study of condensers, you know that paper condensers should have verv large resistances, above 20 meg-. ohms, and electrolytic condensers should have resistance above at least. 250,000 ohms when properly tested. Since filter condenser failure is a very common cause of hum, and excess rectifier tube current, you check these first, then check the electrical values of other parts in the power pack or in the power supply system.

But suppose that the inspection for surface defects gave no sign of the trouble. In this case the Radiotrician would turn to the schematic circuit diagram with one thought in mind: What part of circuit in this particular receiver could become defective and cause hum? As he glances over the power section of the circuit diagram, he may notice a tuned filter; knowing that even small changes in the values of the condenser and choke coil can result in hum, he checks these . parts carefully with his multimeter. Or he may notice a hum bucking coil indicated in the schematic symbol for the loudspeaker; this could well be the cause of the hum, so a careful check of this part is made. There are electrolytic condensers in the power pack, and they have an unfortunate tendency to become leaky with age, ν so he checks them next. There are resistance-capacitance filters in grid circuits for application of a C bias

voltage, so he checks these for leaky condensers and shorted resistors.

But this is not intended to be a lesson telling how to service a radio receiver; we have briefly considered these actual examples of servicing problems primarily to illustrate correct service technique and to show how the schematic circuit diagram and the other information contained in service manuals can be of help in servicing. You are becoming acquainted with the process of figuring or reasoning out probable causes of trouble when certain effects are observed—a process which we call "effect-to-cause reasoning."

A knowledge of how radio circuits work is one essential requirement for successfully applying effect-to-cause reasoning; this knowledge you are now acquiring in your Fundamental Course. Experience with actual radio apparatus and continued reference to schematic circuit diagrams will eventually make effect-to-cause reasoning become second nature to you.

Alignment Procedures. We have by no means exhausted the Radiotrician's uses for circuit diagrams. One highly important use, in connection with the alignment of tuned circuits in a receiver, deserves particular attention, for it is often desirable to realign a repaired receiver to secure maximum selectivity and sensitivity.

In a conventional superheterodyne receiver having peak-tuned circuits, the I.F. stages are aligned first and then the oscillator is made to track the preselector. In an all-wave receiver the preselector-oscillator tracking adjustment must be made for each band. Before any of these adjustments can be made, however, you will first want to know how many trimmers there are to be adjusted, where each one is located, and what it does. A study of the schematic circuit diagram will give you this information. Furthermore, this analysis of the diagram will reveal if there are any image rejection circuits, harmonic or code interference traps, A.F.C. circuit trimmers, or fidelity-equalizing trimmers. Once you spot these on the circuit diagram and note their functions, you can mentally omit them from your alignment procedure and concentrate upon v the true circuit-aligning trimmers. You may note that certain oscillator coils have adjustable iron cores; this establishes the fact that they are the low-frequency oscillator adjustments.

The circuit diagram may reveal the presence of variable-selectivity $I.F.\checkmark$ transformers and band-pass circuits which can be adjusted either for peak response or for broad (band-pass) response. It would be difficult to determine the presence of a band-pass circuit from mere study of the receiver chassis, yet the schematic circuit diagram and your knowledge of circuit operation immediately suggests the correct aligning procedure. If you know radio theory and the usual design practices, there will be no need for you to follow detailed alignment instructions as given in service manuals.

Once you have determined what the correct alignment procedure should be, you must locate the trimmer condenser adjusting screws or nuts on the actual chassis; again the schematic circuit diagram can be of great help in identifying each trimmer. If a pictorial layout is available, do not overlook it during an alignment procedure; this diagram often reveals the exact locations of the various trimmers.

Subjects for Further Study. The preceding brief outline of the technique of radio servicing has shown you many uses for circuit diagrams and the associated information prepared by receiver manufacturers in the form of service manuals. We will now consider in some detail how these service manuals can be used to best advantage in radio servicing, and how the information needed for servicing can be obtained directly from the receiver chassis when no service manual is available. The important subjects for further study are therefore:

- 1. How to identify the various stages on a schematic circuit diagram.
- 2. How to analyze each stage of a schematic circuit diagram.
- How to identify the various stages on the actual receiver chassis.
- How to locate various parts on the chassis with a pictorial layout diagram.
- How to locate parts when no pictorial layout is available.
- 6. How to trace each tube electrode circuit with and without the help of a schematic circuit diagram. ✓
- 7. How to locate the alignment trimmer condensers. \checkmark
- 8. How to appraise the performance of a receiver.
- 9. Important differences between schematic diagrams and actual chassis wiring connections. \checkmark

For the purpose of illustrating the various points covered in this lesson. we have selected a popular model of a well known receiver (Philco Model 38-7) and have reproduced, in the center pages of this lesson, the following material pertaining to this receiver: Fig. 1-the schematic circuit diagram as it appears in the service manual of the receiver; Fig. 2-the pictorial layout diagram for parts underneath the chassis, as given in the service manual; Fig. 3-pictorial layout of parts on the top of the chassis. as given in the service manual; Fig. 4-tube socket connections as seen when looking at the bottom of the chassis, and electrode operating voltages; Fig. 5-reproduction of an actual photograph of the top of the chassis, corresponding to the diagram in Fig 3; Fig. 6—reproduction of an actual photograph of the bottom of

the chassis, corresponding to the dia- γ gram in Fig. 2.

For convenience in referring to these diagrams while studying this lesson, it is suggested that you pry up the staples holding this lesson together, carefully remove the inside four pages (13, 14, 15 and 16), and bend back the staples again. When you-have completed your study of this text-book, you can replace these diagrams in the same manner.

Identifying Stages on a Schematic Diagram

When a Radiotrician first takes up a schematic circuit diagram, he makes a general survey before attempting to locate any individual circuits or parts. in order to determine the general lineup of stages in the receiver. When looking at Fig. 1, for example, he would immediately identify the circuit as that of a two-band superheterodyne receiver which is A.C. operated and has a simple tuned input circuit feeding directly into a pentagrid converter tube which acts as oscillator-mixerfirst detector. He notices that this is followed by a single I.F. amplifier stage which feeds into a double-diode detector in which one section acts as the actual demodulator and the other serves as the A.V.C. tube. The diode demodulator feeds into a single audio amplifier stage which in turn feeds a pentode power output amplifier stage. He notes also that the power pack employs a full-wave rectifier tube. An experienced man could make this identification of stages almost at a glance. for familiarity with the tubes and circuits employed enables him to omit a great many of the steps in the reasoning process now to be described.

Circuit diagrams are invariably arranged so they can be read from left to right when tracing the path taken by intelligence signals from the antenna to the loudspeaker. We therefore start at the upper left of the dia-

gram in Fig. 1, where we find a terminal strip having provisions for two antenna connections and one ground connection. Observe that the signals pass through R.F. transformer 1, the secondary winding of which is tuned by one section of a ganged variable condenser. This tuned circuit feeds into a type 6A8G tube having five grids. Although on this particular diagram the designation DET.-OSC. identifies this stage as the detector-oscillator, you will find many circuit diagrams in which only the tube type numbers are given. We know, however, that the first stage must be either an R.F. amplifier stage or the first detector: we trace from the plate of this tube into I.F. transformer 13, and we know that tubes having five grids (pentagrid conveters) are almost invariably used as combination oscillator-mixer-first detector tubes, so we feel safe in iden-V tifving this stage as the mixer-first detector of a superheterodyne circuit.

Another glance at input R.F. transformer 1 reveals that each of its windings is in series with the winding of another R.F. transformer marked 2, V and that there are switches for shorting out one secondary winding. Clearly, then, we have a two-band receiver. Verification of this can be secured by tracing from the first grid of the 6A8G tube, which we know must be serving as the grid of the local oscillator; this goes to two tuning coils, 5 and 6, and one of these can be shorted out with a switch.

The second tube in our receiver lineup, a 6K7G, is fed by the secondary of the first I.F. transformer and feeds into the primary of the second I.F. transformer; it must therefore be an I.F. amplifier stage. We know that this second I.F. transformer (19) must feed either into a second detector or into another I.F. amplifier stage. The tube in question is a 6J5G triode, and triodes are seldom if ever used as I.F. amplifiers. We must examine this circuit more carefully to determine its true function (assuming that the 2ND'DET-A.V.C. designation is not present). Note that the secondary voltage of second I.F. transformer 19 is applied between the grid of this triode and ground, with the cathode of the tube directly grounded. This looks much like an ordinary diode detector circuit, with the control grid of the triode serving as the diode plate. Resistors 20 and 21 are in the circuit, and now we note a volume control potentiometer connected across resistor 21, with condenser 24 in series to block direct currents. The plate and cathode of this triode tube must therefore be serving as the A.V.C. diode, since the plate is also fed by the second I.F. transformer through D.C. blocking condenser 23.

From the movable contact of the volume control we trace our signal path to the grid of a 6K5G triode, which can only be a first audio amplifier stage since it follows a diode second detector. This tube in turn feeds into a pentode type 6F6G tube having a loudspeaker as its plate load, and consequently this is the power output stage.

One glance at the power transformer symbol in the power pack circuit in Fig. 1 is sufficient to identify our circuit as that of an A.C. receiver, for power transformers are never used in universal or D.C. sets.

There are a great many other clues which will prove useful in identifying each stage when you first go over a circuit diagram to get the general lineup, and consequently you will seldom if ever have any difficulty in identifying a stage even though its function is not indicated on the diagram.

Analyzing Each Stage on a Schematic Diagram

When the servicing procedure has advanced to the point where the defective stage is isolated, it is often necessary to refer to the circuit diagram and analyze in detail that particular stage. Since all of the stages in a receiver are subject to failure, you should know how to analyze any one of the stages when necessary. To show how simple this can be when the analysis is made in the proper manner, we will go through the process for each stage in turn in the circuit of Fig. 1.

Oscillator - Mixer - First Detector. Below the sketch of band-changing switch 48 is a note indicating that this switch is shown in the broadcast band position. Comparison of this diagram with the individual switches in the R.F. input and oscillator coil circuits shows that these separate switch symbols are also in the broadcast band position. For the broadcast band setting, then, the short across the primary. of antenna coil 1 makes this coil inactive. Signals picked up by the an- $\sqrt{}$ tenna (the RED terminal) thus pass through the primary of antenna coil \mathcal{D} and through switch contacts A1-A2 to. ground, inducing in the secondary winding an R.F. voltage which is applied through the inactive secondary of antenna coil 1 to the fourth grid of the 6A8G tube. The fact that the secondary of antenna coil 2 has a higher D.C. resistance than the secondary of antenna coil 1 verifies the fact that coil 2 serves for the broadcast band (a higher resistance usually indicates more turns of wire, giving the higher inductance required for the lower-frequency bands)

The only switch in the oscillator circuit, that for contacts A3-A4, is open during broadcast band operation, and consequently oscillator coil 6 acts in series with oscillator grid coil 5 in the oscillator tuned circuit. The feedback voltage required for oscillation enters this circuit through oscillator plate coil 5, which is in the second grid (oscillator plate) circuit of the 6A8G tube. The oscillator section of the gang tuning condenser connects

between the first grid and ground, and consequently is shunted across both coils in the tuned circuit.

For the short-wave setting of the band-change switch, contacts A1-A2are open, making antenna coil 1 effective, and contacts A6-A7 are closed, shorting the secondary of antenna coil 2 and thereby making this transformer ineffective. Switch contacts A3-A4 in the oscillator circuit are connected together, shorting oscillator coil 6 and thus reducing the inductance in the oscillator tuned circuit to the value required for the higher - frequency band.

Condensers 7 and 8 acting with resistor 9 furnish the automatic C bias for the local oscillator. Condenser 7Bis simply an oscillator coupling condenser, which serves to feed A.C. plate current into the tuned grid circuit during broadcast band operation. Condenser 7 also serves as the low-frequency padder in the oscillator circuit for broadcast band operation. Trimmer condenser 7A is the oscillator high-frequency trimmer for the broadcast band, and trimmer condenser 4Bis the oscillator high-frequency trimmer for the short-wave band. There is no oscillator low-frequency padder on the short-wave range. High-frequency trimmer 4A is the only input circuit alignment adjustment; consequently we can assume that the input circuit or preselector is quite broad in frequency response on the short-wave band, and is adjusted only for the broadcast band.

The pictorial diagrams of the antenna and oscillator coils, at the lower left in Fig. 1, give coil connection information which is often helpful in tracing continuity through the coil circuits. The small numbers on these coil sketches correspond to the numbers on the schematic symbols for the corresponding coils.

I.F. Amplifier. We recognize the 6K7G I.F. amplifier tube as a pentode,

and learn from a tube chart that it is a super-control amplifier tube which can be A.V.C.-controlled. Tracing from the control grid or first grid of this tube through the secondary of the first I.F. transformer, we find the circuit to be through one-megohm resistors 15 and 27 to a tap on the voltage divider in the power pack, and then to ground. This means that the tube receives an initial negative C bias from the power pack. From the common connection of resistors 15 and 27 we, note a lead which can be traced to ground through one diode section of the 6J5G tube (that formed by the plate and cathode), which means that the tube also receives an A.V.C. voltage.

There is a third winding on the first I.F. transformer, which places the suppressor grid at D.C. ground potential. The secondary of this I.F. transformer induces in this coil a signal voltage which can produce either regeneration or degeneration, depending upon the manner and upon the phase relationship of the secondary coil current with respect to the coil voltage. This third coil is connected in such a way that whenever the circuit tends to oscillate (regenerate), a degenerating voltage is applied the suppressor; likewise, when the circuit tends to degenerate during off-tune conditions, a regenerating voltage is applied to the sup-The third winding thus pressor. serves to stabilize the circuit during off-tune conditions.

Following the 6K7G tube is the second I.F. transformer, of conventional design. Trimmer condensers <u>13A</u>, <u>13B</u>, <u>19A</u> and <u>19B</u> permit peak adjustments of the two I.F. transformers.

Second Detector. The output voltage of the I.F. amplifier is applied to the grid and cathode of the 6J5G triode, with resistors 20 and 21 serving as the load for this diode detector circuit. Resistor 20 also acts in combination with condensers 19C and $19D_{\bullet}$. as an I.F. filter. Rectified diode current flowing through resistor 21 develops across it an A.F. voltage which is transferred through D.C. blocking condenser 24 to potentiometer 26. Resistor 32 and condenser 38, connected between a tap on this potentiometer and ground, provide automatic bass compensation. Switch 39 has one contact for shorting out condenser 38, when automatic bass compensation is not desired, and other contacts for tone control circuits. The movable contact on volume control potentiometer 26 feeds A.F. signals through condenser 28 to the grid of the 6K5G first audio amplifier tube.

A.V.C. Tube. The output voltage of the I.F. amplifier also acts upon the plate of the 6J5G triode tube. through condenser 23. The grid and cathode of this tube thus serve as one diode, while the plate and cathode form another diode section which acts as the A.V.C. tube. Resistor 27 (tracedown from the plate of the 6J5G tube) serves as the load for this second diode; V the D.C. voltage required for A.V.C. purposes is developed across this resistor. The circuit from resistor 27 to ground is through the 8-ohm and 35ohm sections of voltage divider 43, and consequently the D.C. voltage drop across these sections is applied to the A.V.C. controlled tubes in the form of a constant negative C bias, as well as a delay voltage for the diode A.V.C. tube. The A.V.C. voltage is applied to the two A.V.C.-controlled tubes -(6A8G and 6K7G) through an A.V.C. filter made up of resistor 15 and condenser 3.

First A.F. Amplifier Stage. A tube chart tells us that the 6K5G tube is a high mu triode designed especially for use in resistance - capacitance coupled audio amplifiers. Resistor 36, coupling condenser 34 and grid resistor 35 make up the resistancecapacitance coupling network through (which this tube feeds into the output

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stage. A negative C bias for the first audio tube is obtained from tap 2 on voltage divider 43, with resistor 33and condenser 30 serving as a filter.

Output Stage. According to a tube chart, the 6F6G tube in the output stage is a power amplifier pentode. The entire voltage drop across voltage divider 43 serves as the negative C bias for this tube, and likewise the entire output voltage of the power pack is applied to the plate of the tube through the primary winding ot output transformer 41. Note that condensers 37 and 40 are connected in series between the plate of the output tube and ground, with the common connection of these condensers going to one contact on tone control switch 39. When the switch is set at this contact (at its extreme right-hand position), condenser 40 is shorted out and condenser 37 then by-passes the higher frequency components of the intelligence signal to ground, giving the effect of a boost in bass notes. When switch 39 is at some other setting, the extremely low capacity of condenser 40 (.008 mfd.) makes the reactance of this shunt path to ground so high that there is very little attenuation of any signal frequencies.

The voice coil of the dynamic loudspeaker, marked 42, is connected across the secondary of output transformer 41. The single-loop coil symbol drawn in series with the voice coil but in the opposite direction represents a hum bucking coil.

Power Pack. Obviously we have a full-wave rectifier circuit here, since the two plates of the 5Y4G rectifier tube are connected to the outer ends, 5 and 7, of the high voltage secondary winding on power transformer 46 Center tap 6 on this secondary winding is therefore the negative or B terminal of the power pack; it traces through resistor 43 to the chassis and ground, making terminals 2, 3 and 4 on this resistor increasingly more negative with respect to point 1, the chassis.

Point 3 on the rectifier tube filament winding is the high-voltage terminal of the rectifier system, so we will find the power pack filter network connected between this point and either the B- terminal or ground. The main filter employs condenser input (45), with loudspeaker field coil 44 serving as a choke and condenser 11A serving as the output filter condenser. The symbols indicate that condensers 45 and 11A are of the electrolytic type. The common terminal of field coil 44 and filter condenser 11A is the B+ or high-voltage output terminal of the power pack, and feeds directly to the plate of 6K7G I.F. tube, the 6K5G. audio amplifier tube and the 6F6G output tube through the plate load of each tube.

A separate filter system is employed for the 6A8G pentagrid converter, however; this is made up of filter resistor 12 and electrolytic condenser 11: A voltage divider network (resistors 16 and 22) is connected between the output of this filter and ground, with the higher output voltage being fed through resistor 10 and the oscillator feed-back coil to the second grid (oscillator plate) of the 6A8G tube. The plate of the 6A8G tube receives this same high output voltage through the primary of the first I.F. transformer. A lower voltage is applied from the common junction of resistors 16 and 22 to the screen grids of the 6A8G and 6K7G tubes, with condenser 14 serving as the screen grid by-pass condenser.

Filament connections are indicated but not completed on this diagram. Observe that terminal \mathcal{S} of the filament winding on power transformer $\mathcal{46}$ is grounded and that one filament terminal of each tube except the rectifier is likewise grounded. The other filament terminal for each tube is terminated in an arrow, as also is the For each tube, he first adjusts the tester controls according to a tube chart furnished with the tester, then plugs the tube into a socket on the tester. If the meter in the tester indicates BAD, the tube is defective.

Although tube testers have a number of different sockets and controls, they are really quite simple to use once the operating instructions are carefully studied. There is no need for you to secure a tube tester yet, however—in fact, you will not need one until you are doing a considerable volume of radio service work.

How Tubes Fail. How do radio tubes go bad? Three common reasons are a burned-out filament, low emission, and shorted electrodes.

Burned-Out Filament. The filament in a radio tube is normally operated at red heat or even brighter. The thinnest parts of the filament wire get the hottest. The intense heat eventually causes the filament wire to melt and break open at a thin spot, causing complete failure of the tube.



When the meter pointer in the tube tester does not move at all from its position at the lowest end of the BAD region, the serviceman knows

the tube has a burned-out filament.

Low Emission. Another common cause of radio tube failure is aging of the chemical coating on the **cathode**. This chemical coating gives off electrons when heated by the filament. As the coating wears off during use, fewer and fewer electrons are given off, and the total electron flow from cathode to plate through the tube is correspondingly reduced. Eventually, the number of electrons given off or emitted is so much reduced that the tube can no longer do its full job in the receiver. Radio men then say that the tube has *low emission*. A tube tester would indicate that the tube was BAD.

When a serviceman encounters a

FIG. 10. Tubes are tested by radio men with tube testers like this.



tube which has low emission, he discards it and puts in a new tube.

Shorted Electrodes. Another type of radio tube failure is shorted electrodes inside a tube. Changes in temperature may cause an electrode to sag out of position enough so that it touches a nearby electrode inside the tube. We then say that the tube has shorted electrodes. Thus, the filament may touch the surrounding cathode and cause a short.

Practically all tube testers have a separate red light on the panel of the instrument. This red light glows whenever any two of the electrodes are shorted together. When this light glows, the radio man knows that he will have to replace the tube anyway, so he ordinarily does not bother to figure out which electrodes are touching.

More About Tubes. You will learn a lot more about radio tubes and their electrodes later in the course in a lesson devoted entirely to radio tubes. Furthermore, in one of the N.R.I. Practical Job Sheets, you will get complete step-by-step instructions for testing tubes with typical modern tube testers. other filament winding terminal; this of course, indicates that all these terminals are connected together. All tube filaments except the rectifier are thus connected together in parallel across the filament winding having terminals 8 and θ .

Observe that two condensers, marked 47, are connected in series across the 110 volt A.C. power line, with their midpoints grounded. These condensers shunt to ground any noise interference signals which might otherwise enter the receiver.

The extreme left position of switch 39 is the off or open position for the receiver on-off switch which is in series with the primary of the power trans-This power switch is closed former. for all other positions of switch 39. Moving this switch one contact to the right shorts out automatic bass compensation condenser 38, providing normal receiver operation. Moving the switch one more contact to the right places this condenser in the circuit. and the final position of the switch to the right shorts out condenser 40, giving a bass-boosting effect together with automatic bass compensation.

A note at the bottom of the schematic circuit diagram indicates that on some receiver models employing this circuit, a shadowgraph indicator is inserted in series with the plate supply lead to the I.F. amplifier tube, with a .05 mfd. condenser connected between the plate side of this meter and the chassis to keep R.F. current out of the meter. The small diagram at the lower right of the circuit diagram indicates that pilot lamp 49 is connected in parallel with the tube filaments across the filament winding on the power transformer. A note indicates that an extra pilot lamp (49X)is used on one model to operate a shadowgraph tuning indicator.

Remember that it will seldom if ever be necessary for you to analyze a complete receiver circuit as we have just done. Modern servicing techniques isolate the trouble to a particular stage or section, making it necessary to analyze that one small part of the receiver circuit. Furthermore, do not expect to be able to analyze a diagram or even a part of a diagram completely right from the start. As you acquire additional knowledge and experience, you will find it easier and easier to secure the information which schematic diagrams can give you.

Identifying Stages on the Chassis

By including top-of-the-chassis layout diagrams along with the schematic diagrams in some service manuals, receiver manufacturers have made it quite easy to identify each tube and its function. A typical diagram of this type, in which the position of each tube is clearly indicated with respect to easily recognizable parts such as the power transformer and the tuning condenser, is shown in Fig. 3. \checkmark With this layout at hand, it is a simple matter to perform either a dynamic stage-by-stage elimination test or a circuit disturbance test when hunting for the defective stage.

A tube layout diagram has other uses in radio servicing. Since octalbase tubes are used in practically all modern receivers, any tube will fit into any socket. The sockets on the chassis are ordinarily not marked for the proper tubes, and consequently the tube layout diagram serves as a valuable. guide for replacing tubes when all are removed for testing, or in order to clean the chassis. The wise Radiotrician usually does not depend upon a tube layout diagram, however; he removes only one tube at a time, returning it or replacing it with a new tube before testing the next tube.

But how do we go about identifying the stages on the chassis when no schematic diagram and no tube layout diagram are available? All we



FIG. 5. Top of chassis of Philco model 38-7 receiver



FIG. 1. Schematic circuit diagram of Philco model 38-7 two-band superheterodyne receiver. A dot at the intersection of circuit lines indicates a connection; when two circuit lines cross and there is no dot at the intersection, there is no connection between them. NOTE: For convenience, lift up the staples and remove these four center pages of illustrations temporarily while studying this book.






FIG. 3. Pictorial layout diagram for parts above chassis of Philco model 38-7 receiver.



FIG. 4. Tube socket connection diagram and electrode voltages for Philco model 38-7 receiver.

Scn	em,	Part
N	o. Description	No.
1	Antenna Transformer-Short Wave	32-2558
2	Antenna Transformer-Broadcast	32-2557
3	Condenser .05 mf.	30-4519
4	Tuning Condenser, Models 8 and 9	31-2026
	Tuning Condenser, Model 7	31-2040
5	Osc. Transformer-Short Wave	32-2560
6	Osc. Transformer-Broadcast	32-2559
7	Compensator Dual Models 8 and 9	31-6188
-	Compensator, 580 KC, (Model 7)	31-6195
7.4	Compensator Model 7 (1500 KC)	31-6196
8	Condenser 3500 mmf, mica.	30-1094
9	Resistor 70 000 ohms (14 watt)	33-370339
10	Resistor 5000 ohms (14 watt)	33-250330
11	Condenser, Electrolytic Dual 14 and 8	00-200000
	mfd)	30.2217
12	Resistor 10 000 ohms (3 watt)	33.310630
13	lat I F Transformer	32-2580
14	Condensor 1 mf	30 4455
15	Registor 1 0 mag (14 wett)	33.510330
16	Resistor 10 000 (1 watt)	22 210420
17	Condenses Of mf (29 9 ania)	20 4454
10	Shadammatar (28 8 anlu)	45 9207
10	and I E Transformer	20 22007
20	Desister \$1.000 (mental in 10)	32-2382
20	Resistor 51,000 (mounted in 19)	33-331338
60	Resistor 490,000 onms ('2 watt)	33-449338
22	Resistor 51,000 onms (1 watt)	33-351435
23	Condenser, mica, 110 mmf.	30-1031
24	Condenser .01 mi.	30-4479
25	Removed Prior to Production	
26	Volume Control	33-5216
27	Resistor 1 meg. (1/2 watt)	33-510339
28	Condenser .015 mf.	30-4358
29	Audio Shorting Switch (38-7 only) Part	
	of Selector Crank	
30	Condenser .1 mf	30-4499
31	Resistor 1.0 meg. (1/2 watt).	33-510339
32	Resistor 51,000 (1/2 watt)	33-351339
33	Resistor 1.0 meg. (1/2 watt)	33-510339
34	Condenser .015 mf	30-4515
35	Resistor 1.0 meg. (1/2 watt)	33-510339
36	Resistor 99,000 (14 watt)	33-399339
37	Condenser .03 mf	30-4447
38	Condenser .06 mf.	30-4467
39	Tone Control.	42-1327
40	Condenser .008 mf	30-4112
41	Output Transformer (Model 7)	32-7862
	Output Transformer (Models 8 and 9) .	32-7019
42	Cone and Voice Coil Assembly (H31)	36-3801
	Conc and Voice Coil Assembly (K41)	36-3174
	Cone and Voice Coil Assembly (HS)	36-3796
	Cone and Voice Coil Assembly (S7)	36-3157
43	Bias Resistor	33-3316
44	Field Coil Assembly (H31).	36-3665
	Field Coil Assembly (K41)	36-3931
	Field Coil Assembly (HS)	36-3690
	Field Coil Assembly (S7)	36-3039
45	Electrolytic Condenser	30-2219
46	Power Transformer, 115V, 50/60 evele	32-7833
	Power Transformer, 110V, 25 to 40 cycle	32-7627
	Power Transformer, 115/230V 50/60	
	evela	32-7835
47	Condenser 015-015 mf 25 mf	3703 DG
48	Wave Switch	42-1325
49	Pilot Lamp, Models 8 and 9	34-2064
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have is the chassis itself, which can be represented in this discussion by the photographs in Figs. 5 and 6. We will concentrate upon Fig. δ , which shows a top view of the chassis, since you will often want to identify the tubes and stages without removing the chassis from its cabinet.

The power transformer and the two-gang tuning condenser are, of course, easily identified. Likewise you can be pretty sure that the two cylindrical metal cans adjacent to the tuning condenser, each having two adjusting screws, are I.F. transformers. Their presence identifies the receiver as a superheterodyne. The fact that there is a lead coming out of one I.F. transformer to the top cap of a tube is further proof that we are _ dealing with a superheterodyne, for in tuned radio frequency receivers the top caps of tubes usually go to the stator sections of the gang tuning condenser.

The shapes of the completely enclosed metal cans, one to the right and below the power transformer, and the other at the lower right on the chassis, identify these parts as electrolytic filter condensers; with experience you will be able to spot these. at a glance.

Once you have determined the type, of circuit, you can proceed to identify the tubes. First of all, look for the rectifier tube, which will have two rectangular electrodes with a heavy filament wire inside each, but no other electrodes. We notice a tube like this, marked 5Y4G, at the upper right of the power transformer (see Fig. δ ; reference to a tube chart verifies that this is a rectifier tube.

Below the rectifier tube and next to the power transformer in Fig. 5 is a tube which is larger than all other tubes except the rectifier. This has a rather large plate, with a cathode and several other elements inside. By counting the elements or by determ- $\sqrt{}$ the tube chart tells us definitely that

ining its number and referring to a tube chart, we identify this as a pentode; since there is no top cap connection for the control grid, we know definitely that it is not an R.F. pentode. It is logical to assume, then, that we have a power output pentode, a tube widely used in the output. stages of receivers. A tube chart would verify this assumption.

To the left of the tuning condenser in Fig. 5 is a tube marked 6A8G; a tube chart identifies this as a pentagrid converter, the fourth grid of which connects to the top cap. We say, therefore, that this tube is for the oscillator - mixer - first detector stage. A lead from the top cap of this tube goes to the upper stator section of the tuning condenser, identifying it as the preselector tuning condenser. The other section of the tuning condenser must therefore tune the oscillator circuit.

At the right of the rectifier tube in . Fig. 5 is a tube marked 6K7G; according to our tube chart, this is an R.F. pentode. Since there are only two sections in the gang tuning condenser, one for the first detector input and one for the oscillator, we know that there is no R.F. amplifier stage: This means that the 6K7G R.F. pentode must be used as an I.F. amplifier: the fact that there is a lead coming from the I.F. transformer to the top cap of this tube verifies our as-The tube chart also resumption. veals that this is a super-control tube, and since a receiver designer ordinarily does not employ such a tube unless it is to be A.V.C.-controlled. we know that this receiver has A.V.C.

The two tubes at the upper right on the chassis in Fig. 5 still remain to be identified. The upper is marked 6J5G, and reference to a tube chart reveals it to be a triode which can have a number of different uses. The lower tube is marked 6K5G but here

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it is a high-gain audio amplifier tube. Since we have not yet located the second detector-A.V.C. tube, with only these two tubes left to identify, we can say that the 6K5G is the first audio amplifier tube and the 6J5G is the second detector-A.V.C. tube.-We could verify this by measuring the voltage between the plate and cathode of each tube; we would find that the tube selected as the audio amplifier had a reasonably high D.C. supply voltage, whereas the plate of the 6J5G tube was slightly negative with respect to the cathode. These measurements would tell us that a delay voltage was present in the A.V.C. system.

We have just seen how each tube in this receiver can be identified as to its function without removing the chassis from its cabinet. Naturally a person would not go through this procedure if a tube diagram were at hand, or if a schematic circuit diagram were available, for it is not always possible to determine the function of each tube on a chassis without tracing connections underneath the chassis. On the other hand, with experience you will be able to identify the various tube stages on most receivers almost at a glance, and will not have to refer to diagrams.

General Suggestions. The ability to recognize radio parts by their appearance alone is obviously quite desirable when it is necessary to identify stages on the chassis. For this reason it is important that you keep in touch with new developments as announced in radio magazines and in the latest catalogs of mail order radio supply houses or of your local radio parts distributor.

You can generally secure the latest tube charts from your local parts distributor, or from the manufacturer of the tubes which you handle. These charts will familiarize you with the appearances and characteristics of new and old tubes alike.

The first thing you want to know, when working on a chassis without a service manual, is whether you have a T.R.F. or superheterodyne receiver. Look for I.F. transformers, as they are clues identifying a superheterodyne; if these I.F. transformers are under the chassis, as they often are, look for the screws of their adjusting trimmers on the top of the chassis. If one of the sections of the gang tuning condenser has rotor plates which are shaped differently from those in other sections, you have another clue toward a superheterodyne. A receiver having only two sections in the gang tuning condenser but with six or more tubes can reasonably be assumed to be a superheterodyne. Older radio receivers with three, four and five sections in the condenser gang will generally be T.R.F. sets, but make sure that there are no I. F. transformers before deciding this.

Next you will want to know how the receiver is powered. If there is a multi-lead cable in place of the line cord, it is a battery receiver; if there is a two-wire cord having at its end the familiar two-prong wall outlet plug, you can be sure that it is a socket-powered receiver. But do we have an A. C., D. C., or universal socket-powered receiver? If a power transformer is visible on the chassis, you can be sure it is an A. C. receiver. If there is a rectifier tube but no power transformer, it is a universal A.C.-D.C. receiver. If there is no power transformer and no rectifier tube, it is a D.C. receiver. These rules will enable you to identify practically all receivers, but be on the lookout for exceptions. A few transformerless A. C. receivers have been made; these will have the conventional full-wave rectifier tube, so you will have to check the rectifier connections to make a positive identification. *

Now let us concentrate our attention upon the stages ahead of the I.F. amplifier in a superheterodyne receiver. A two-section ganged variable condenser means that there is no R.F. amplifier stage ahead of the first detector, for one section serves for tuning the oscillator and the other for tuning the input to the first detector. Three sections ordinarily indicate that there is one stage of R.F. amplification ahead of the mixer-first detector, but occasionally you may find a bandpass preselector instead. You may find a pentagrid converter tube or a pentode tube being used as a combination mixer-first detector-oscillator, or there may be a pentode serving as mixer-first detector and a separate triode (or even a screen grid or pentode tube) serving as the oscillator tube.

Most superheterodynes use screen grid or pentode tubes in the R.F. first detector and I.F. stages; this means that the control grid connection for each of these tubes will be to a top cap. In the case of a pentagrid converter, the top cap will be connected to the fourth grid inside the tube, and there will be an external connection from this cap to the stator section of the detector input tuning condenser;

* There is seldom need to make a positive identification unless you isolate a defect to the power pack. In this case you will naturally remove the chassis, and an inspection of rectifier connections will reveal the type of circuit used. If it is an A.C. set with a voltage-doubling power pack circuit, you will usually find that the plate of one diode section is directly connected to the cathode of the other diode section. There will be two electrolytic condensers connected in series, with the negative lead going to the remaining diode plate and the positive lead to the remaining cathode. The common connection of these two condensers goes to one side of the power line, and the other side of the power line goes to the The common plate-cathode connection. circuit for this is given in an earlier lesson in your Course.

with a three-gang condenser the oscillator stator will be at one end, and will not have a connection to a tube cap if a pentagrid converter is used.

Because I.F. transformers require complete shielding, they will be found in metal shields or cans and will usually be mounted on the top of the chassis. The secondary of the first I.F. transformer will always be connected to the control grid of the I. F. amplifier tube; since this will be a screen grid or pentode tube, you can expect to find a flexible lead coming out of the top or side of the I.F. transformer and going to the top cap of the tube.

When there are two I.F. transformers, the second will feed into the second detector, usually by means of an under-chassis connection (only when the second detector is a screen grid or pentode tube will there be a connection from the second I.F. transformer to its top cap).

Since diodes or triodes are ordinarily used as second detectors, you can usually assume that any tube having a top cap connection into an I.F. transformer is an I.F. amplifier tube. If there are three I. F. transformers, look for two I.F. amplifier tubes; in all probability you will find them in line with the mixer-first detector and the second detector.

The second detector has a few peculiarities which permit easy identification. Look for a double diode, a double diode-triode, a double diodepentode, or a triode following the last I.F. transformer. Occasionally you may run into a screen grid or pentode second detector which will have a lead from its top cap to the last I.F. transformer. When the R.F. and I.F. stages use pentode tubes and there is a pentagrid converter in the line-up, look for tubes such as the 56, 6C5, 6C6, 55, 85, 75, 6B8 and 6H6 serving as the second detector, the second detector-A.V.C. tube, or as the second detector-A.V.C.-first A.F. amplifier.

Rectifier tubes in an A.C. receiver are easy to locate, for they are almost always right next to the power transformer. Obviously it would not be good design practice to run high-voltage A.C. leads any distance through the chassis from the power transformer to the rectifier tube. Look for such tubes as the 80, 25Z5, 82, 83 5Z3, 5Z4 and 6X5, for these are all rectifier tubes. Most of these have glass envelopes, through which you can usually see two long black rectangular or cylindrical anodes, with a thick Vshaped filament inside each.

Output tubes can be recognized by their numbers and by their relatively large size in comparison to all other tubes except the rectifier. Doubletriode power amplifier tubes such as the 19, 6N7, 6A6, 79 and 53 are instantly identified as output tubes connected for push-pull or push-push operation. Such tubes as the 45, 47, 2A3, 6L6, 2A5, 59, 43 and 42 are power output tubes which you will find used singly or in pairs for push-pull or push-push operation.

An additional tube, usually a triode, located between the second detector and the power tubes or near them is very likely an audio voltage amplifier tube.

Cathode ray tuning indicator tubes can, of course, be identified on sight. Automatic frequency control tubes, noise-limiting and noise-suppression tubes, A.V.C. tubes and A.V.C. amplifier tubes are not so easy to identify from the top of the chassis. When the circuit contains extra tubes which are not readily identified, it is wiser to secure the circuit diagram and identify these extra tubes by a process of elimination or by comparing tube numbers. With some receivers it is practically impossible to identify all tubes merely by studying the top of

the chassis; in these cases it is either necessary to refer to a service manual or remove the chassis and trace tube connections. When metal tubes are used, the tube numbers stamped on the sides of the tubes must serve as your guide.

Locating Parts with a Pictorial Layout Diagram

Let us suppose that the receiver being used as our example in this lesson has developed an annoving squeal. We might logically suspect that a screen grid by-pass condenser is open somewhere; referring to the schematic circuit diagram in Fig. 1, we see that screen grid by-pass condenser 14 for the 6K7G I.F. amplifier tube is a likely offender. We can make a quick check by shunting this condenser with a good unit of the correct value (.1 mfd.), provided that we can locate condenser 14. An examination of the underside of the chassis reveals at least nine tubular paper condensers which could be 14; by referring to the parts layout diagram in Fig. 2, however, we learn that the condenser under suspicion is near the back edge of the chassis and is between two tube sockets. With this information, it is easily located on the chassis; turn to Fig. 6 and see how quickly you can locate it on the photo. If the squeal stops when we temporarily shunt this condenser with a good condenser, we know that condenser 14 requires replacement.

Now let us consider the case where the receiver has an annoying hum, and an inspection for surface defects reveals that the rectifier tube is gassy (a blue glow can be seen between its electrodes). Before inserting a new tube, we naturally want to check the electrolytic condensers with an ohmmeter to determine if excessive leakage through them was the cause of rectifier tube failure. To locate these condensers, we first refer to the circuit diagram and determine what numbers are assigned to them. At the right of the rectifier tube in Fig. 1 are two electrolytic condensers; 45 and 11A, while above the power transformer is another, marked 11. This numbering indicates that 11 and 11A are in the same housing, and the fact that only two electrolytic condensers were found on top of the chassis is further proof of this assumption.

We locate condenser 11 on the pictorial layout diagram in Fig. 2, finding it in the upper right corner of the diagram, near the volume control. An inspection of the connections underneath this chassis would show three leads, colored black, green and red respectively. This means that the two condenser sections must have a common connection: an inspection of the schematic circuit diagram verifies this and shows that the common negative terminal is grounded. The black lead on the condenser itself is grounded, so this must be negative A the red and green leads are therefore the positive leads for these electrolytic condensers. A leakage test is now readily made with an ohmmeter. connecting the plus terminal of the ohmmeter (that terminal which goes to the plus terminal of the ohmmeter battery) to the plus terminal of the The practical man selcondenser. dom bothers to figure out condenser polarity, however; he connects both ways and uses the highest-resistance reading, for he knows he will then have the correct polarity. One of the condenser leads is disconnected for. this test, as there are usually other parts shunting the condenser.

Let us consider one more case, that where the receiver is dead and the defect is localized to the pentagrid converter by means of a stage-bystage elimination test. Measurements of electrode voltages on this tube show that there are no D.C. supply voltages: other measurements reveal that there is no screen grid voltage on the I.F. amplifier tube. A study of the schematic circuit diagram reveals that an open circuit through resistor 12 could be a cause of the trouble. The parts layout diagram in Fig. 2 shows that resistor 12 is located un. der the chassis, between electrolytic. condenser 45 and I.F. transformer 13-Since it is a 10,000 ohm resistor, we have the additional clue that it will have a brown body, black end and orange dot in accordance with the RMA color code for resistors. You should now be able to locate this resistor in Fig. 6; it is a large carbon resistor, supported at one end by a metal clamp under which is white insulating paper. Naturally you would look for a defect in this resistor or for a break in its connection.

These two examples show the value of the parts layout diagram in bridging the gap between the schematic circuit diagram and the actual receiver chassis.

Locating Parts without a Pictorial Layout Diagram

The procedure for locating on the chassis a part which is indicated on the schematic circuit diagram becomes considerably more involved when no pictorial layout diagram is available. Familiarity with the appearance of various radio parts will speed your search, as also will experience in tracing actual wiring on a chassis. A few examples will illustrate the procedure to be followed.

On the schematic circuit diagram, condenser 14 is shown connected between the screen grid of the 6K7G I.F. amplifier tube and ground; let us see how we would go about locating this condenser on the chassis. First of all we locate the 6K7G tube

Above-Chassis Parts

Although other superheterodyne radio receivers may look quite different from the average receiver chosen for our example in this lesson, all superheterodyne receivers use the same fundamental principles of operation and similar groups of ordinary radio parts. Corresponding parts in superheterodyne receivers may look different, but their important purposes and actions will be quite similar.

The Loudspeaker

Let us consider the loudspeaker next, for it is the largest radio part in our receiver, and has the actual job of changing the final strong audio signal into the sound waves of a radio program.

Electrodynamic Loudspeaker. As you previously learned, a modern dynamic loudspeaker has only three important parts: A diaphragm or cone, a voice coil, and a magnet.

In the loudspeaker used in our superheterodyne receiver, an electromagnet provides the magnetic field, and consequently the loudspeaker is known as an *electrodynamic loudspeaker*. The coil of the electromagnet is called the *field coil*.

The direct current required by the field coil is obtained from the power pack of the receiver. An electrodynamic loudspeaker will therefore have two extra wires, going from the field coil to the proper power pack terminals underneath the chassis. The field coil and its leads can be seen in Fig. 11, which shows the loudspeaker all by itself on the chassis of our receiver.

Through its connection to the power pack, the field coil also does an important filtering job, about which you will learn more later when you study power packs.

How It Works. When an audio signal current is sent through the voice coil of our electrodynamic loudspeaker, this voice coil produces a varying magnetic field which reacts with the constant magnetic field of the field coil.

The two coils thus alternately attract and repel each other. The field coil electromagnet cannot move because it is attached to the frame, so



FIG, 11. Here's a close-up view of the loudspeaker in our superhet receiver.

only the voice coil moves back and forth. Since the voice coil is attached to the cone, the cone likewise moves the surrounding air back and forth, producing the sound waves of radio programs.

Damaged Cone. When the stiff paper cone of a loudspeaker is accidentally torn, or when it becomes warped or cracked due to aging, radio programs no longer will sound natural, and will sometimes be accompanied by buzzing or scratchy sounds.

When a Radiotrician encounters a damaged cone, he either replaces the

on the top of the chassis and determine the position of its socket underneath the chassis. We would find this socket to be at the rear edge of the chassis; almost exactly midway between the sides of the chassis (you can locate it for yourself on Fig. 6 by referring to the socket connection diagram in Fig. 4).

The next step is the location of the screen grid terminal on this socket; a tube chart will give this information if a socket connection diagram like that in Fig. 4 is not available. We look for terminal S, for this letter (as well as the notations G_{a} or G_2) are used to designate the screen grid terminal. We know that condenser 14 is connected between this terminal and ground, and a careful study of the schematic diagram shows that there is no other condenser connected to this screen grid terminal. We trace each lead in turn from this terminal of the socket until we locate one going to a tubular paper condenser marked .1 mfd.; we note that the other lead of this condenser goes to a soldering lug which is riveted to the chassis, and thus have definite proof that this is condenser 14.

Here is another example: suppose that the trouble has been isolated to the R.F. input circuit. We suspect an open circuit in one of the windings of R.F. transformers 1 and 2. A study of the schematic circuit diagram shows that if we connect an ohmmeter between the two antennà terminals marked RED and BLK. we can check continuity of the primary winding. If the ohmmeter indicates a resistance of about .1 ohms when the band change switch is in the broadcast position (shorting contacts A1 and A2), we know that the primary of R.F. transformer 2 is good. A changeover to the short-wave setting of the switch removes the short across the primary of R.F. transformer 1, and if the ohmmeter reads about .2 ohm, we know that this primary winding also is good.

Now we study the circuit diagram again to see how we can check the continuity of the secondary coils. VObserve that there is a complete conductive path from the fourth grid of the 6A8G tube to the secondary of the first I.F. transformer and control grid (top cap) of the 6K7G I.F. amplifier tube. The diagram further indicates that if we place an ohmmeter between the top caps of these two tubes, we should measure a D.C. resistance of .1+2.3+14, or a total of approximately 16.4 ohms. We do not even have to remove the chassis from the cabinet in order to make this test.

If, when measuring the resistance between the top caps of the first two tubes, we secured an infinite resistance reading, we would know that an open circuit existed somewhere along the path being checked. To make sure that the trouble is an R.F. coil, we could make an ohmmeter test between the top cap of the 6K7G tube and the chassis. The schematic circuit diagram indicates a resistance of a little over 2 megohms (the combined resistance of parts 15, 27 and a section of 43) between these two points. Any resistance reading differing greatly from this value, or an open circuit reading, would indicate trouble in the I.F. transformer. If the ohmmeter reads the correct value of 2 megohms. we make the same test between the top cap of the 6A8G tube and the chassis; let us assume that we get an open circuit reading.

To determine which of the R.F. coils is defective, we must locate them underneath the chassis and test each one individually. The sketches at the lower left in Fig. 1, with each coil connection numbered and the relative lengths of the windings shown, makes identification of these coils on the chassis quite simple.

Oftentimes, however, these coil pictures are not provided on the circuit diagram. In this event we would trace from the top cap of the 6A8G tube to the input tuning condenser stator, and from there to point 3 on the secondary of R.F. transformer 1. (Each stator section has two terminals, one above and one below; the R.F. transformers are underneath the chassis, so we look underneath the chassis for a lead coming through it directly under the stator section in question.) We trace through the heavy wire forming the secondary winding to the other terminal, marked. 4, and then in turn trace to point A6on the band change switch, to point 3of the secondary of R.F. transformer 2, and through this winding to terminal 4. We can then make a continuity test across this entire circuit or any part of it.

Tracing Tube Electrode Circuits

Once the various tube stages have been identified, the defective stage isolated and the tube in that stage checked, the next logical step is a check-up of the circuits in the defective stage. These tests are simple and easy to make if you recognize that electrode circuits in any receiver can always be traced to certain definite points. It is then a simple matter to make continuity tests of the paths between the tube electrodes and these points. Let us trace the conductive paths from the electrodes in Fig. 1 before considering the general continuity-checking rules which apply to all circuits.

Plate Circuits. Suppose we start at the plate of the 6K7G I.F. amplifier tube and trace a conductive path as far as we can. From the plate we go through the primary of second I.F. transformer 19 in Fig. 1, down a lead

marked *RED* to loudspeaker field coil 44, and through this coil to the filament of the 5Y4G rectifier tube. Clearly the rectifier filament is one terminal in this path.

Now trace from the plate of the 6A8G pentagrid converter tube through the primary of first I.F. transformer 13 and down another lead marked *RED* through resistor 12 to the rectifier filament. The plate of the 6K5G first audio amplifier tube traces' through resistor 36 and loudspeaker field coil 44 to the rectifier filament, and likewise the plate of the 6F6G output tube traces through the primary of the output transformer and through field coil winding 44 to the rectifier filament.

There is a very good reason why the plates of all these tubes should trace to the rectifier filament or to a rectifier cathode. In any rectifier type of power pack the filament is the highest positivo voltage terminal, and consequently all electrodes which are supplied with a positive voltage must eventually trace to the rectifier filament.

Screen Grid Circuits. Since the screen grid of a tube is likewise supplied with a positive D.C. potential, you should also expect to trace a conductive path from it to the rectifier filament. We can check this very easily in Fig. 1. From the screen grid of the 6K7G tube (the middle grid, which is also connected directly to the screen grid of the 6A8G detector-oscillator tube), we trace through resistors 16 and 12 to the rectifier filament. In a similar manner we can trace from the screen grid of the 6F6G output tube through loudspeaker field coil 44 to the rectifier, filament.

The only other electrode in this particular circuit which requires a positive D.C. potential is the second grid of the 6A8G pentagrid converter

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tube, which serves as the anode of the oscillator. Observe that this traces through oscillator plate coil 5 and then <u>through resistors 10 and 12 to</u> the rectifier filament.

Control Grid Circuits. Since the control grid of an amplifier tube must be negatively biased with respect to its cathode, we know that there must be a conductive path between the control grid and the cathode of an amplifier tube. Let us verify this on the circuit in Fig. 1. The fourth grid of the 6A8G tube is serving as a control grid, so we trace from it through the secondary winding of R.F. transformer 1 and 2 and then through resistors 15, 27 and 43 to the chassis and the grounded cathode. Clearly there is a continuous path here for direct current. The first grid of this tube, serving as the oscillator control grid, likewise traces to the cathode through oscillator grid coil 5, through coil 6 and then through resistor 9 and the chassis. The control grid of the 6F6G output tube traces through resistors 35 and 43 to the chassis and cathode. A circuit can likewise be traced from the other control grids to their respective cathodes through the chassis.

As you know, the center tap of the high voltage secondary winding on the power transformer (feeding the two plates of the rectifier tube) is the lowest negative D.C. terminal in the power supply. A further study of Fig. 1 will show that the control grids can also be traced to this most negative point in the power pack. This means that in an A.C. receiver you can, after turning the set off, make a continuity test of each control grid circuit by connecting one ohmmeter lead to the control grid and the other to the cathode of the tube, to one of the plates of the rectifier tube or to the most negative point in the power pack. In a universal receiver, where there is no power transformer, the most negative D.C. point will be the receiver side of the main power switch; be sure this switch is open when you make your test.

Suppressor Grid Circuits. When an external prong connection is provided for the suppressor grid of a tube, this will either trace to the cathode of the tube or to some negative supply terminal. For continuity checking purposes, then, you may treat the suppressor grid just as if it were a control grid.

Diode Detector Circuits. Since a diode detector is a rectifier and consequently passes direct current, we know that there must be a conductive path from the plate through the external circuit to the cathode. An ohmmeter connected between the cathode and plate of a diode detector should therefore indicate continuity. (There may be an exception to this rule in the case of a power pack rectifier tube used as a diode rectifier, for the load on this diode may be tubes in the receiver which are conductive only when their cathodes are heated.)

The second detector-A.V.C. tube in Fig. 1 employs two diodes, one for detection and one for A.V.C. purposes, but the plate of each can be traced to the cathode. For example, the actual plate of the tube traces through resistors 27 and 43 to the chassis and then to the cathode, while the grid (which serves as the other diode plate) traces through the secondary of I.F. transformer 19 and through resistors 20 and 21 to the chassis and cathode.

General Rules for Checking Continuity of Electrode Circuits in A.C. Receivers

1. There should be a conductive path from the rectifier tube filament or cathode (the highest positive D.C.

terminal) to all tube electrodes which are supplied with a positive D.C. potential, such as plates and screen grids.

- 2. There should be a conductive path from the most negative D.C. terminal in the power pack to all control grids and suppressor grids which require zero or negative bias voltages. Likewise there should be a conductive path between the control grid and the cathode of a tube, and between the suppressor grid and cathode of a tube.
- 3. In diode detectors or diode A.V.C. tubes, there should be a conductive path between the plate and the cathode.

Failure to secure a conductive path in any of the cases mentioned, as evidenced by an infinite-resistance reading of the ohmmeter, indicates a break in the electrode circuit in question. It should then be a simple matter to check the circuit piece by piece to locate the break.

When the schematic circuit diagram and the parts layout diagram are available, all parts suspected of being open can be located first on these diagrams and then on the chassis. Each part is then checked for continuity, making sure that the test includes the connecting leads.

When neither the circuit diagram nor the pictorial layout diagram is available, the circuit must be traced on the chassis for the most likely path between the points in question before making ohmmeter measurements. For example, if you found that there was no continuity between the plate of the 6K5G first audio amplifier tube and the rectifier filament, you would first locate the plate terminal of the audio tube socket. There would be a wire from this terminal running around the front edge of the chassis to a terminal to which a resistor and a .015 mfd. condenser are soldered. Of course, no continuity should be expected through the condenser, so you trace through the resistor to the screen grid of the

power output tube and then to the field coil of the loudspeaker. You know that your circuit should trace through this field coil rather than through leads going to the electrodes of other tubes and so you arrive at the rectifier filament. Having traced the circuit, you can then proceed to test continuity of each part and section of it.

How to Locate Trimmer Condensers

As you know, the location and identification of each trimmer condenser in a receiver is essential for alignment purposes. If the manufacturer supplies a circuit diagram like that in Fig. 1 and a trimmer layout diagram like that in Fig. 3, this task is simple. When no trimmer layout diagram is available, however, the problem becomes more complicated. Let us see how this would be done on the receiver being used as an example if the only service manual data available is the circuit diagram in Fig. 1.

First of all, we note that there are two holes in the top of each I.F. transformer through which can be seen adjusting screws. Those in the first I.F. transformer we can identify as I.F. trimmers 13A and 13B, while those in the second I.F. transformer must logically be I.F. trimmers 19A and 19B. We next note that there is a trimmer condenser mounted on each stator section of the gang tuning condenser. That one which is on the R.F. input stator section we identify as high-frequency trimmer 4A; the trimmer on the oscillator stator section must therefore be oscillatorhigh frequency trimmer 4B.

A glance over the circuit diagram in Fig. 1 shows that there are only two more trimmer condensers, 7 and 7A. Since manufacturers will make all trimmer condenser adjustments available from the top of the chassis wherever possible, we look for these on the chassis and finally locate the two adjusting screws to the right and a little above the gang tuning condenser unit, near the last I.F. transformer. We must turn the chassis over and trace connections, however, in order to tell which is 7 and which is 7A. Figure 1 shows that these trimmers have one common connection through condenser 8 to ground, with the other connections going to opposite ends of oscillator coil 6. Trimmer 7A also connects to one terminal on oscillator coil 5, so we can positively identify as trimmer 7A on the chassis that trimmer which connects to two oscillator coils. The remaining trimmer, which connects to only one oscillator coil, will therefore be 7.

General Trimmer-Locating Suggestions. For each band in an all-wave superheterodyne receiver there will usually be the following separate trimmers: 1, one high - frequency trimmer for each preselector stage which is tuned to the incoming R. F. carrier frequency; 2, a high-frequency oscillator trimmer; 3, a lowfrequency oscillator padder. · (The receiver used as an example in this lesson is an exception to this rule, for there is one high-frequency oscillator trimmer for each band but only a lowfrequency oscillator padder for the broadcast band. In the preselector, one high-frequency trimmer serves both bands.) In the case of a three-band receiver having one R.F. amplifier stage, there would be four trimmers in the preselector and oscillator sections for each band, making a total of twelve trimmers ahead of the I.F. amplifier. Usually there will be two trimmers for each I.F. transformer. but in the less expensive receivers which have no R.F. amplification and only one stage of I.F. amplification, a high gain I.F. transformer using

only one tuned circuit and consequently only one trimmer may be found.

Look for I.F. trimmer adjustment screws either at the top or on the side of the I.F. transformer shield; occasionally, however, you may find them at the bottom of the transformer, visible only from the bottom of the chassis. Some manufacturers mount these trimmers directly on the chassis rather than in the shield, but they will always be located close to the I.F. transformers they tune.

In a conventional broadcast band superheterodyne receiver, all of the high frequency trimmer condensers will be mounted on the stator plates of the condensers which they adjust. The low-frequency padder will be a separate unit, mounted somewhere near the oscillator coil or the oscillator tuning condenser section. If two or more trimmers are found on the chassis of a broadcast band receiver, trace the circuit to each in order to make sure that the one selected as the padder is not a wave trap adjustment or some other special circuit arrangement. The oscillator low-frequency padder will generally be in series with the oscillator coil; sometimes this padder will be shunted by a fixed condenser.

Ordinarily there will be no highfrequency trimmers mounted on the gang tuning condenser of an all-wave receiver; these will be located near the coils which they adjust, instead. (The Philco two-band receiver used as our example in the lesson is not considered an all-wave receiver.) Sometimes all of the oscillator circuit coils will be found in a separate aluminum shield can, along with the oscillator high-frequency trimmers, while another shield can will be used for the preselector coils feeding into the mixer-first detector, and one more shield for the antenna coils if an R.F.

amplifier stage is used. There will be one adjusting screw on each of these shields for each band in the receiver; for example, in the case of a three-band receiver there will be three high-frequency trimmer adjusting screws on the side or on top of each preselector and oscillator coil shield.

Sometimes the preselector and oscillator coils are placed below the chassis, completely shielded from each other and from other parts by metal partitions and a metal cover plate. The adjusting screws for the high-frequency trimmers will be accessible through holes in the cover plate.

Low-frequency padder condensers are ordinarily mounted on the chassis, and are usually ganged or grouped together on a common insulating strip. Identification of these in the case of an all-wave receiver can sometimes be made by noting the number of plates in each unit; that padder having the largest number of plates will be for the broadcast band, while the other with the fewest plates will be for the high-frequency band. It is best, however, to identify trimmers by referring to a circuit diagram and to a chassis layout when available. Experienced Radiotricians can often identify the bands controlled by preselector and oscillator trimmers simply by adjusting each in turn and noting the effects; if you attempt to do this, however, be sure to note the original setting of the trimmer so you can restore this setting in case the wrong trimmer is chosen.

Appraising Receiver Performance

When servicing a receiver, it is obviously a waste of time to attempt to secure better performance than was originally intended by the manufacturer. Each receiver is designed to sell for a definite price, and consequently, lower-priced receivers will lack many of the features which are incorporated in the more expensive sets.

Actual experience with the various types of receivers is, of course, the best guide for determining when a receiver is performing in a satisfactory manner, but there are a number of recognizable clues indicating how much can be expected from a particular receiver in the way of selectivity, sensitivity and fidelity.

First of all, the number of tuned stages in a receiver is an excellent guide as to its selectivity and sensitivity, for increasing the number of tuned stages improves both of these performance characteristics.

The presence of more than two tuned circuits between adjacent R.F. or I.F. amplifier tubes indicates improved selectivity, possibly with some sacrifice in gain (sensitivity). Plate and grid connections to taps on tuning coils are signs of improved selectivity; the use of these taps also indicates that the stages in the receiver provide more gain than is considered essential, for the introduction of a selectivity-improving scheme tends to cut down the gain.

The number of tubes in the entire R.F. and I.F. amplifier systems, not including the oscillator and oscillator control tubes, is another rough guide to the amount of selectivity and sensitivity in a receiver. A superheterodyne will give better selectivity and better gain than a T.R.F. receiver having an equal number of tubes, simply because the I.F. amplifier in a super utilizes higher Q factor circuits and more tuned circuits per stage than does a T.R.F. receiver. Furthermore the gain and selectivity will be more uniform over the entire band in a super than in a T.R.F. receiver.

The presence of a stage of R. F. amplification ahead of the mixerfirst detector in a super indicates a highly sensitive receiver with a high ratio of signal level to converter noise level. In other words, with an amplifier stage in_the preselector,_weak_ stations should be heard with clarity and freedom from converter noise interference.

Do not expect the selectivity and gain of an all-wave receiver to be as good on the higher-frequency bands as on the broadcast band, for the Q factors of the tuning coils are considerably reduced at the higher frequencies.

When checking the selectivity of a receiver, pay no attention to the space on the tuning dial which is covered by a station, but rather, note whether stations which are received at your location with approximately equal intensity can be heard separately even though only 10 kc. apart.

To determine what you can expect from a particular receiver in the way of fidelity, study the loudspeaker and the compartment in back of it carefully. If the design of these indicates a wide range of frequency response, means for suppressing cavity resonance and means for utilizing bass reflection, you can be quite sure that high fidelity performance was intended.

The presence of band-pass circuits in the R.F. and I.F. amplifiers, especially in the preselector circuits, are equally important clues pointing to a high fidelity receiver. The mere fact that I.F. transformers have a tuned primary and tuned secondary is no indication that they are band-passed, but if these transformers are critically coupled or over-coupled (as indicated by your inability to secure a sharp peak response curve with a cathode ray oscilloscope), they are very likely designed for band-pass

use. If a receiver employs three or more I.F. transformers, and the design of the preselector appears to indicate a broad response, then bandpassing is probably present.

An I.F. amplifier circuit employing variable_coupling between the windings of the I.F. transformer is a fidelity control, permitting the customer to choose between high fidelity and high selectivity.

In receivers having automatic frequency control, the regular I.F. amplifier circuits are very likely intended to be band-passed for high fidelity if the discriminator is fed through one or more independent and highly selective tuned circuits.

Schematic and Chassis Wiring

The fact that the actual wiring on a chassis may be considerably different from the connections indicated on the schematic circuit diagram cannot be emphasized too strongly even though the differences between schematic and actual chassis wiring have already been taken up in this Course. Remember that the lines on a schematic diagram merely indicate which terminals of the various parts are connected together *electrically*; these lines are not intended to show how wires actually run from point to point on the chassis, nor do they indicate the relative lengths of the connecting wires.

For example, referring to the first audio amplifier stage in Fig. 1, notice that parts 34 and 36 are both connected to the plate of this first audio tube. It might be possible to solder the condenser and resistor leads directly to the plate terminal of this tube socket. Actually, however, the resistor is soldered directly to this tube terminal and a long lead run from this terminal to condenser 34, which is soldered directly to the grid terminal of the tube socket in the following stage. Experience with actual receivers will show you that other electrical connections indicated on a schematic circuit diagram can be made in many different ways on the actual chassis.

TEST QUESTIONS

Be sure to number your Answer Sheet 32FR-2.

Place your Student Number on every Answer Sheet.

Never hold up one set of lesson answers until you have another ready to send in. Send each lesson in by itself before you start on the next lesson. In this way we will be able to work together much more closely, you'll get more out of your Course, and you will receive the best possible lesson service.

- 1. Is it necessary for a Radiotrician to analyze the complete circuit diagram of a receiver when repairing a defect? $\mathcal{N}\mathcal{D}$
- 2. Into what two general trouble groups can all defective radio receivers be placed? DEAD RECEIVER PLAYS TMPROPERLY.
- 3. What is the simplest and speediest test for isolating the defective stage in a dead receiver? CIRCHIT DISTURBANCE TEST,
- Referring to the schematic circuit diagram in Fig. 1, what symbol identifies the circuit as that of an A.C. receiver. POWER TRANSFORMER.
- 5. Give the numbers of the trimmer condensers in Fig. 1 which permit peak adjustments of the two I.F. transformers. /3A, /3B, /9A, $\neq /90$,
- 6. Near what easily identified part would you expect to find the power pack rectifier tube in an A.C. receiver? BY TRANSFORMER.
- 7. Through what parts does the second grid (oscillator plate) of the 6A8G tube in Fig. 1 trace to the rectifier filament? R-#10 = 5000 --8. What is the general rule for checking continuity in the circuits of all
- 8. What is the general rule for checking continuity in the circuits of all electrodes (in an A.C. receiver) which are supplied with a positive D.C. potential?
- 9. Should there be a conductive path (continuity) between the plate and such as serve
- 10. Are high frequency trimmer condensers ordinarily mounted on the gang tuning condenser of an all-wave receiver?

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PERSONA'L NEATNESS PAYS

Clothes may not make the man, but all too often they do determine the degree of his success in business. The radio man who keeps himself clean and neat gets the respect of his customers, and that one thing pays big dividends.

Good appearance is achieved by wearing appropriate conservative clothes for whatever kind of work you are doing, and keeping these clothes clean and well pressed at all times. Thus, inexpensive brown work shirts and trousers are fine for shop use and even for service calls, *provided they are clean and pressed*. But keep those shoes shined, keep your hair combed and cut, wash your hands immediately after handling a dusty or greasy chassis, and watch all those other little personal-neatness items which are being stressed so much in advertisements today.

To sum up, the cost of maintaining a neat personal appearance is negligible in comparison to the higher price a neat, well-dressed, business like radio man can demand and get for his services. Neatness inspires confidence, and confidence in turn makes customers accept your prices without question. Make a habit of neatness—it pays!

J. E. SMITH

BEHAVIOR OF RADIO WAVES AND RECEIVING ANTENNAS

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entire cone-voice coil assembly with an exact duplicate replacement assembly obtained from the set manufacturer or a radio parts distributor.



or installs an entirely new loudspeaker. Curiously, with small loudspeakers it is often cheaper and more satisfactory to put in a new loud-

speaker than to replace or repair a damaged cone.

Off-Center Voice Coil. One common loudspeaker trouble is an offcenter voice coil. This coil moves in and out in a very small air gap between the pole pieces. Slight warping of the cone or slight shifting of the various parts of the loudspeaker throw the coil off center in the air gap, and the coil then rubs against the pole pieces.

When the voice coil is off center, radio programs will sound distorted on the loud bass notes which are present when a man talks or sings, but will probably be perfectly clear for women's voices, which have few low notes. You may hear a rubbing sound when you push the cone in and out with your fingers after the set is turned off.

When the voice coil is properly centered, you should be able to move it in or out without having the coil rub against the pole pieces between which it moves.

Practically all of the older loudspeakers and even a great many modern loudspeakers have provisions for recentering the voice coil. These loudspeakers will either have one screw inside the voice coil or several screws arranged around the outside of the voice coil. When these screws are loosened, the voice coil can be moved a small distance sidewise in any direction for recentering purposes. With the screws loose, small narrow strips of celluloid or cardboard are inserted between the inside of the voice coil and the central iron core of the loudspeaker to provide automatic centering. The screws are then tightened, after which the centering strips can be removed.

There is a growing trend toward permanent fastening of the voice coil to the loudspeaker frame. The loudspeaker used in our superheterodyne receiver is an example of this, and its voice coil cannot economically be recentered by the radio serviceman. If for any reason the voice coil gets off center, the loudspeaker can be returned to a loudspeaker cone manufacturer or a radio supply firm for installation of a new cone-voice coil unit, or the entire loudspeaker can be replaced.

Open Field Coil. A broken wire (usually due to corrosion) is about the only trouble encountered in the field coil of an average electrodynamic loudspeaker. The trouble usually occurs at one end of the coil, at the point where the flexible insulated field coil lead is connected to the enamel-covered coil wire. If the trouble is at the outer end of the coil, it can often be repaired after removing the outer layer of insulating cloth from the coil.

When the field coil opens up and the break cannot be repaired, it is sometimes possible to remove the defective field coil and install a new one. In many of the smaller modern loudspeakers, however, the core of the field coil is permanently anchored in position during manufacture, making it impossible to remove the field coil. The entire loudspeaker must then be replaced.

Replacing a Loudspeaker. After a receiver has been in use for three or four years, the paper used in its cone

STUDY SCHEDULE No. 33

For each study step, read the assigned pages first at your usual speed, then reread slowly one or more times. Finish with one quick reading to fix the important facts firmly in your mind, then answer the Lesson Questions specified for that step. Study each other step in this same way.

1. Introduction; Radio Waves; Radiation Patterns of Transmit-

ting Antennas; Vertical and Horizontal Polarization Pages 1-6 Components of a Radio Wave; How Radio Waves Are Produced; Wavelength; Wavelength of an Antenna; Frequency-Wavelength Relationship; Field Intensity Surveys; How Radiation Patterns Are Secured; Action of a Receiving Antenna; Polarization of Radio Waves. Answer Lesson Questions 1, 2 and 3.

- 3. Receiving Antenna Problems; Directivity of Receiving

Three factors in selecting and installing a receiving antenna; four factors affecting directional characteristics of receiving antennas; Current and Voltage Distribution of Half-Wave Doublet Antennas; Current and Voltage Distribution of Grounded Vertical Antennas; Directional Characteristics; Radiation Patterns for Receiving Antennas; summary of receiving antenna installation rules. Answer Lesson Questions 6 and 7.

- 5. All-Wave Antennas; Variable-Directivity Antennas Pages 24-29 Types of All-Wave Antennas Commonly Used; Multiple Doublet Antennas; Single All-Wave Doublets; Use of Counterpoise for Noise Reduction; antennas with controllable directional characteristics.
- 6. Mail your Answers for this Lesson to N.R.I. for Grading.
- 7. Start Studying the Next Lesson.

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Behavior of Radio Waves and Receiving Antennas

Introduction

THE extremely high sensitivity of modern radio receivers has made many servicemen somewhat careless with the installation of receiving antennas. In addition to feeding strong signals from desired stations into a receiver, a good receiving antenna must also have a high signal-to-noise ratio, if station programs are to be heard with a minimum of interfering noises. The introduction and rapid acceptance of all-wave receivers brought added responsibilities to the serviceman, along with puzzling questions concerning the peculiarities of radio waves at various frequencies.

In order to answer these highly practical questions and solve the receiving antenna problems which occur in service work, you must first understand what radio waves are, how they behave, how a receiving antenna can be made to intercept a maximum of radiated energy, and how intercepted signals can be fed to a receiver with highest possible efficiency. All this information will be given you in this lesson.

Radio Waves

Components of a Radio Wave. Whenever electrons flow through a conductor, both an electric field and a magnetic field will always be produced. (Of course, either field can exist by itself if no electron motion is involved; an electric field exists alone between two electrically charged bodies, while a magnetic field exists around a permanent magnet.) Furthermore, whenever electrons change their speed of travel through a conductor or oscillate back and forth, the result is a moving electric field and a moving magnetic field which together are known as an electromagnetic field. If the electrons are oscillating back and forth at a high enough rate (at a high frequency), the electromagnetic field will travel off into space at the speed of light (186,000 miles per second), giving what is known as an electromagnetic wave or a radio wave. A radio wave, therefore, has two components: 1. The electric field component; 2. The magnetic field component.

How Radio Waves Are Produced. A study of the fields in the vicinity of a typical transmitting antenna such as that shown in Fig. 1A will give you a good idea of how radio waves are radiated into space. This is a doublet antenna, for it is broken at its center and the two sections are fed with an oscillating electron flow produced by a radio transmitter; it is also known as a dipole antenna.* Suppose that the transmitter polarity at one instant of time is as indicated in Fig. 1A; electron flow will then be in the directions indicated by arrows *i*, making the upper end of the doublet negative and the lower end positive. We can consider the circuit between the two halves of the antenna to be completed through space, for

^{*}Although you will find the terms "doublet antenna" and "di-pole antenna" used interchangeably in many cases, the following definitions are accepted by most radio engineers: A doublet antenna is fed at its center, and its two sections are essentially equal in length and are in the same straight line; a doublet antenna may or may not be at resonance for the frequency of the signal fed to it. A di-pole antenna is to the radio physicist a theoretically perfect antenna having uniform current throughout its length and with ends having opposite polarity; to the practical radio man a di-pole antenna has essentially uniform current throughout its length, is not at resonance with the signal frequency fed to it, has ends with opposite polarity, and may be fed anywhere along its length.

these halves form a condenser having the surrounding air as a dielectric. A *displacement current* (the equivalent of an electron movement) flows from the negative half of the antenna through space to the positive half.

To simplify our study, we can neglect the transmitter connection and consider this antenna as a single length of wire in space, having the polarity shown in Fig. 1B when electron flow iis upward as indicated; this antenna wire can be thought of as a source which is feeding power into space. The polarity will reverse from instant to instant of time in step with reversals in electron flow through the wire.

When electron flow through a doublet antenna is in the direction shown in Fig. 1B, a magnetic field will encircle the wire and will act in the direction shown in Fig. 1C; this magnetic field is usually designated by the letter H. At the same time an electric field having the direction shown in Fig. 1D will be set up in space by the oppositely charged halves of the antenna; this electric field is usually designated by the letter E. At any point in space, the electric and magnetic fields (E and H) produced by an energized antenna will always be at right angles to each other.

A continuously oscillating (alternating) electron flow in a doublet antenna produces in the vicinity of the antenna an electromagnetic field; the relative intensities of the electric and magnetic components of this field at various distances away from the antenna at one instant of time are shown in Fig. 1E. At point 1, for example, there is an electric field component E_1 which is acting upward parallel to the antenna (in the plane of this page), and a magnetic field component H_1 which is at right angles to the electric field (towards you, at right angles to this page). The arrow lines E_1 and H_1 are intended to show the relative strengths of the electric and magnetic

fields at point 1, and the directions in which these fields act; these arrow lines do not represent the paths of electric and magnetic fields, for these paths are shown in Figs. 1C and 1D. Each other point in Fig. 1E, such as 2, 3, 4, 5, 6, 7 and 8, will likewise have its electromagnetic field with an electric component E and a magnetic component H; note that between points 4 and 8 $\mathbf{8}$ there is a reversal in the directions of the components, with E now acting downward in the plane of the paper and H acting away from you perpendicular to the paper. The electromagnetic field is traveling away from the transmitting antenna in all directions, at the speed of light; both the electric and the magnetic components of this field are always acting at right angles to this direction of travel.

A portion of the electromagnetic field produced by a transmitting antenna acts as if it were permanently associated with the antenna and were staying in its immediate vicinity; this portion is known as the induction field. The other part of the electromagnetic field, known as the radiation field, the electromagnetic wave or the radio wave, breaks away from the antenna and travels outward through space. Insofar as energy is concerned, the E and H components of the radiation field or radio wave are always equal to each other at a distance from a transmitting antenna (in the region of the radiation field); close to the antenna, in the region of the induction field, the E component may be considerably greater than the H component. We will consider only the radiation field (radio) wave) for the remainder of this lesson unless otherwise indicated, since the induction field ordinarily exists only for a few hundred feet away from the antenna.

Wavelength. During the time required for one complete cycle of the alternating current flowing through the antenna in Fig. 1E, the electromagnetic wave will travel from point 0 to point 8, and both the electric field E and the magnetic field H at any point will also go through one complete sine wave cycle. The distance between point 0 and 8 is known as one wavelength.

When engineers study radio waves in space, they ordinarily consider only one component (usually the electric field component E), and consequently they think of a radio wave as having the simple sine wave form shown in Fig. 1F. The distance between the extremities of one complete cycle of this wave is therefore one wavelength. Fractions of a cycle are called fractions of a wavelength; half a cycle is thus

Wavelength of an Antenna. The characteristics of a receiving antenna depend to a great extent upon its length with respect to the wavelength of the signal being picked up. For this reason, it is common practice to describe an antenna as being "so many" wavelengths long for a particular signal frequency. When a radio signal produces a half-sine-wave distribution of current along a receiving antenna, we know that the antenna is half as long as one wavelength of that particular signal, and we say that the antenna is one-half wavelength long.

We could easily determine the wavelength of an antenna at a particular frequency by drawing the radio wave



FIG. 1. These diagrams show the first steps in the formation of a radio wave by a simple doublet antenna which is fed by a transmitter. Electron flow i through the antenna wire in the direction shown produces a magnetic field H and an electric field E, both of which travel away from the antenna at the speed of light.

half a wavelength or simply a halfwave, and $1\frac{1}{2}$ cycles are $1\frac{1}{2}$ wavelengths or 3/2 waves.

The next important fact for you to recognize is that a wavelength is a different value (in feet or in meters) at each frequency. Doubling the frequency means that we get twice as many alternations or cycles of the radio wave into a given distance, and each cycle or wave will therefore be only one-half as long; Fig. 1G illustrates this clearly. The higher the frequency of a radio wave, the shorter will be the length of one complete wave or cycle. and the antenna on the same line at the same scale, and simply counting the number of cycles covered by the antenna. Figure 1G illustrates this; line CD, which represents an antenna drawn to scale, covers $1\frac{1}{2}$ cycles and consequently we have a 3/2-wave antenna at this particular signal frequency. It is generally more convenient to figure out this information than to secure it graphically, however; simply divide the length of the antenna by the wavelength of the signal (using the same units of length), and the result will be the wavelength of the antenna. Example: Antenna length is 100 feet: signal wavelength is 200 feet; $100 \div 200 = \frac{1}{2}$, and consequently we have a half-wave antenna in this example.

Frequency - Wavelength Relationship. Oftentimes the frequency of a radio wave rather than the wavelength is known. The preceding section indicates that there is a definite relationship between frequency and wavelength; this relationship is based upon the fact that an electromagnetic wave in space always travels at the same speed, approximately 300,000,000 meters per second or 186,000 miles per second (this is also the speed at which light travels through space). The following formula gives the relationship between the speed, frequency and wavelength of a radio wave in space:

Wavelength in meters multiplied by frequency in cycles per second always equals 300,000,000.*

Figure 1E shows only one of the many possible paths for radio waves away from the antenna. There will be many other paths in all directions. When a radio wave travels along the ground directly to the receiving antenna, it is known as a ground wave. When a radio wave travels up into the sky and is then bent back to the receiving antenna, it is known as a sky

To find frequency when wavelength is known, use this variation: f = 300,000,000+ λ . Example: If wavelength is 5 meters, what is the frequency? Answer: f = 300,-000,000 + 5 = 60,000,000 cycles or 60 megacycles.

To find wavelength when frequency is known, use this variation: $\lambda = 300,000,000$ $\div f$. Example: If frequency is 1,500 kc., what is the wavelength? Answer $\lambda = 300,-000,000 \div 1,500,000$, or 200 meters.

To get wavelength in feet, multiply the wavelength in meters by 3.28 (roughly 3¼). wave. Both the ground wave and the sky wave must be considered when studying problems involving the transmission and reception of radio signals.

Radiation Patterns of Transmitting Antennas

It is a well-known fact that radio stations which are intended to serve a local area, such as broadcast band stations in this country, use antennas which radiate signals having maximum intensity along the ground. Stations which must send messages or programs to far-distant receiving points use antennas which direct radio waves skyward, for it is now known that there is a layer in the upper atmosphere which will bend the sky wave back to earth again at a point considerably distant from the station. Commercial short-wave stations in transoceanic use, amateur radio stations, and shortwave stations which transmit entertainment programs to far-distant countries all use antennas which are carefully designed to produce the strongest possible sky wave.

Field Intensity Surveys. Careful measurements must be made of the amount of energy radiated in each direction (north, south, east and west) from a transmitting station in order tc determine whether an antenna is giving adequate coverage of the desired local service area or is radiating satisfactory signals to desired distant points, as the case may be. It is customary to measure the energy in the electrical component E of the radio wave; the results are expressed in terms of microvolts per meter, millivolts per meter or volts per meter, with these units expressing the voltage existing between two points which are exactly one meter apart in space at the location where the signal intensity is measured. These field intensity units also correspond to the voltage which would be measured between the ends

^{*}When dealing with antennas it is usually more convenient to speak in terms of wavelength than frequency. If either the frequency or the wavelength of a radio wave is known, you can easily determine the missing value. It is common practice to let f represent frequency in cycles per second and λ (Greek letter lambda) represent wavelength in meters; the above formula then becomes $\lambda \ge f = 300,000,000$.

of a true di-pole antenna exactly one meter long, erected at the receiving location being checked. Measurements are usually made with a loop receiving antenna, and the readings are converted by means of higher mathematics to equivalent values for a true di-pole antenna. The final values are plotted on a graph to give what is known as the radiation pattern of a transmitting antenna.

How Radiation Patterns Are Secured. If an engineer wants to know how well the ground wave is getting out in various directions from a broadcast station antenna, he might take suitable field intensity measuring equipment first to a position P_1 , which is two miles due east of the antenna, place his pick-up antenna at the angle and direction which give maximum signal, and make the first measurement; he could then travel in a twomile-radius circle around the antenna and make measurements at positions P_2 , P_3 , P_4 , etc., in Fig. 2A, which are NE, N, NW, W, etc., from the antenna along this circle (any other convenient radius could just as well be used).

An alternative procedure involves selecting convenient positions each two miles away, regardless of their direction from the transmitting antenna, and measuring the angle θ (see Fig. 2A) between a line to the antenna and a line running due east from the antenna. After measurements have been made for a sufficient number of points, the engineer draws a diagram similar to that in Fig. 2A, with radial lines running from the antenna location to each point at which a measurement was made. Along line $A-P_1$, starting from A, he plots the signal intensity measured at point P_1 , using any convenient scale (he might let one inch represent 10 millivolts per meter). The length of line AE_1 then represents the field intensity measured at point P_1 , two miles due east of the antenna. The same procedure is repeated for each other radial line, and a smooth curve is drawn through the various points. This curve is the horizontal radiation pat-L tern; it indicates the approximate signal intensity in millivolts per meter which can be expected at a point two miles from the transmitting antenna at any angle with the due east line, eliminating the need for making measurements at all angles. The particular pattern shown in Fig. 2A indicates that practically no energy is being radiated along the ground in the NW, W and SW directions, and that almost uniform energy is being radiated in the N, NE, E, SE, and S directions.



. Horizontal and vertical radiation patterns for a directional transmitting antenna. FIG. 2.

Similar information concerning sky wave radiation can be obtained by making measurements from an airplane or blimp. An example of a pattern secured for a flight due east and west over an antenna is shown in Fig. 2B. This radiation pattern indicates that signal intensity is a maximum along the ground, but that considerable energy is being radiated into the sky in the form of a sky wave.

Vertical and Horizontal Polarization

Action of a Receiving Antenna. A short straight antenna will pick up a maximum signal when it is parallel to the electric lines of force associated with a radio wave, as indicated in Fig. 3A. Free electrons in the antenna wire move back and forth along the wire in step with changes in the direction and intensity of the electric lines of force, with the result that an alternating current is sent down to the receiver. When the antenna is at right angles to the electric lines of force, the electrons will oscillate from side to side across the wire, and no current will flow to the receiver.

You can also visualize the action of a receiving antenna by considering the magnetic lines of force, which are at right angles to the antenna. These magnetic lines, moving through space at the speed of light, cut across the receiving antenna and induce in it a radio frequency voltage which is proportional to the number of lines which cut the antenna per unit of time. This voltage in turn forces a current down the antenna lead-in wire to the receiver.

Polarization of Radio Waves. When the electric lines of force associated with a radio wave are perpendicular to



FIG. 3. The electric field *B* associated with a radio wave may have any angle whatsoever with respect to ground when it strives at a receiving antenna, and this angle will have an important effect upon signal pick-up by a given antenna. When *E* is perpendicular to the ground, we say that the radio wave is vertically polarized; when *E* is parallel to the ground, the radio wave is horizontally polarized.

the ground, as indicated in Fig. 3B, they are said to be vertically polarized. When the electric lines of force associated with a radio wave are parallel to the ground, as indicated in Fig. 3C, they are said to be horizontally polarized. The magnetic lines of force associated with a radio wave are also polarized, but we can neglect them when considering the effectiveness of various receiving antennas.

The electric lines of force associated with a sky wave seldom have perfect vertical or horizontal polarization; they are usually at an angle with the ground, somewhat as shown in Fig. 4. Here the radio wave is traveling toward the ground after reflection from

the sky. The intensity and direction of the electric field at point P (above the ground on this path of travel) can be represented by the arrow line E. This arrow line can be considered to have two components; a vertical component $E_{\rm v}$ and a horizontal component $E_{\rm m}$ as shown in Fig. 4. To secure a maximum signal at location P, you would have to place a straight doublet antenna in the direction indicated by E. A vertical antenna would only pick up a signal corresponding to the length of $E_{\rm v}$, and a horizontal antenna would in this case give but slightly more signal pick-up corresponding to the length of $E_{\rm H}$.

The Ground Wave

The transmitting antennas used by broadcast stations for local coverage are generally of the vertical type, for



FIG. 4. An electric field arriving at the angle θ to ground may be considered equal to a vertically polarized component $E_{\rm V}$ acting with a horizontally polarized component $E_{\rm H}$.

these can be designed to radiate considerable energy along the ground and at the same time keep the sky wave at a low enough intensity to avoid interference betwen ground and sky waves within the service area of the station. The electric lines of force produced by an energized vertical antenna are essentially parallel to the antenna when near the surface of the earth, and hence are very nearly vertically polarized. We might conclude that for maximum signal pick-up at the receiving location, the receiving antenna should be of the vertical type, but we know that the majority of receiving antennas are actually horizontal.

There are several reasons why a

horizontal antenna is more satisfactory for receiving purposes. First of all, high vertical antennas are expensive and difficult to erect, making them unsuited for the average home. Secondly, electric lines of force bend forward as they move away from the antenna. as indicated in Fig. 5B, due to the presence of the ground; one engineer describes this phenomenon by saving "the radio waves are pushed forward with their feet dragging on the ground." Another reason is this: Radio waves which travel over hilly or mountainous country or near tall steel buildings are reflected enough to change their angle of polarization and their direction of travel. At a reasonably distant feceiving location such as that shown in Fig. 5A, there is invariably a large enough horizontal component $E_{\rm H}$ (Fig. 5B) to excite a horizontal antenna. 1.

There is still another good reason for making receiving antennas horizontal. Electrical devices which cause manmade interference produce far more vertically polarized electric lines of force than horizontally polarized lines, and since these interference-producing devices are close to the receiver location, they would induce strong signals in a vertical receiving antenna. A horizontal antenna picks up a minimum of noise signals and still intercepts enough of the signal from a local transmitting antenna to give a satisfactorily high signal-to-noise ratio. In most cases, a horizontal antenna gives better reception of sky wayes than does a vertical antenna.

Ground waves radiated by either a broadcast or short-wave station will follow the earth until they are totally absorbed or are too weak to affect receiving antennas. Only by increasing the amount of power radiated by the transmitting antenna can stronger ground waves and more distant reception of ground waves be secured. All reception at points outside the range of ground waves will be due to sky waves reflected back to earth by a layer in the upper atmosphere. These sky waves are absorbed very little by the atmosphere, and consequently will often be reflected back to points thousands of miles away from the station with surprisingly high signal strength.

Television Receiving Antenna Problems

At the carrier frequencies used in modern television systems, automobile ignition systems and electrical devices having sparking contacts can cause considerable interference. It is highly desirable to use a horizontal receiving



FIG. 5. Vertically polarized electric lines of force produced by a vertical transmitting antenna gradually change their angle with the ground as they travel away from the antenna.

antenna in order to keep out these vertically polarized interference signals. Television transmitting antennas are usually designed to radiate horizontally polarized radio waves, for these will be picked up with maximum efficiency by a horizontal receiving antenna and will therefore give a maximum signalto-noise ratio.

Television signals may reach a receiving antenna in a city location over two or more different paths. This multiple transmission effect is due to reflection of waves from buildings, bridges and other steel structures, and may give two or more images superimposed on each other due to slight differences in the times of arrival of the signals over the various paths. This secondary or ghost image effect can be minimized by erecting the horizontal doublet receiving antenna as high as possible and in a direction which gives maximum /reception of either the direct wave or the reflected wave, but not both. When there are two television stations in a locality, a doublet antenna must necessarily be in a position which gives a satisfactory compromise between signals arriving from the two directions. An alternative procedure involves the use of two doublets at right angles to each other and crossed at their centers, with the system oriented to give best possible pick-up from both stations.

Beam Characteristics of Ultra-High-Frequency Waves. Ultra - high - frequency (u.h.f.) radio waves such as are used in modern television systems are above 40 megacycles. These u.h.f. waves behave like beams of light, in that they travel in essentially straight



FIG. 6. Radio waves at ultra-high frequencies travel in essentially straight lines, limiting the reliable range of reception of television programs to lineof-sight distances.

lines from the transmitting antenna. This means that if the transmitting antenna is not visible from the top of the receiving antenna on a clear day, signals will in all probability not be received. With antennas at ordinary heights on level ground, the curvature of the earth definitely restricts the coverage of a given station. This is illustrated in exaggerated form in Fig. 6A; increasing the distance between the two antennas would make the radio waves pass over the top of the receiving antenna without affecting it, for these u.h.f. waves cannot bend around the earth. Television broadcasts from the tower of the highest building in the world, the Empire State Building in New York City, can normally be \checkmark picked up only within a radius of about sixty miles of the transmitting station. Occasionally, reception at greater distances is possible because of a peculiar distribution of moisture in the air

which bends ultra-high-frequency sky waves back to the earth at more distant locations. In Fig. 6B, for example, reception of signals over the direct path marked P_1 would be blocked by the earth, but under certain conditions the sky waves might be bont back to-the lower atmosphere, taking path P_2 to the receiving antenna. In general, however, sky wave reception at ultra-high frequencies is unreliable.

Sky Waves; The Kennelly-Heaviside Layer

Before Marconi in 1901 succeeded ir. transmitting a radio message from England to the North American Continent, scientists generally believed that all radio reception was by means of ground waves, and that long-distance transmission was impossible because of the absorption of energy by the ground. The success of Marconi proved this theory false, and many new theories were advanced to explain the amazing long-distance transmission of signals. Kennelly in America and Heaviside in England simultaneously presented the explanation that there are electron lavers in the sky which act like any conductive layer in reflecting and bending electromagnetic waves. These theories have been proved correct, and today we speak of this wavebending layer in the atmosphere as the Kennelly-Heaviside layer. The reason for the existence of this layer is usually explained in the following manner: The surface of the earth is surrounded by air, which thins out (becomes rarefied) away from the earth. Under the action of ultra-violet rays of light from the sun, this rarefied air becomes ionized, with the result that a layer made up of free electrons, positive ions and negative ions is formed high above the earth.

A radio wave passing into the Kennelly-Heaviside layer will impart some of its energy to the free electrons in the

layer, setting them into vibration. The velocity of vibration will be the least at the highest frequencies, for even an electron has a certain amount of mass which prevents large amplitudes of vibration at high frequencies. The mass of the electron makes it act as a reactance, so its velocity is 90° out of phase with the velocity of the radio wave. The oscillating electrons in the layer produce a new radio wave which acts with the original wave. This combining of velocities which are 90° out of phase results in an apparent speeding up of the radio wave when it hits the electron laver.

Radio waves at low radio frequencies cause the electrons in the layer to vibrate with great amplitude. These electrons collide with air particles, giving up their energy and liberating ions and more electrons. In this way a portion of the energy in the original radio wave is lost in the Kennelly-Heaviside layer; the loss may amount to as much as 30% and occurs chiefly at the lower levels, where there are more ions of air to absorb energy.

Refraction of Radio Waves. Let us assume that the Kennelly-Heaviside layer has a uniform distribution of electrons and that the electric lines of force associated with a radio wave are moving toward it at the angle shown in Fig. 7A. As the radio wave enters the layer, portion y encounters electrons first and speeds up. The result is a bending of the electric lines of force and of the radio wave path itself; this bending of the path of travel is always down toward the earth. Scientists refer to this phenomenon as refraction rather than as bending.

Progressive Refraction. The electron density (number of electrons per unit volume) in the Kennelly - Heaviside layer is greater in the upper regions than in the lower regions. We would naturally expect this, since the upper portion of the layer is closer to the sun and is consequently subject to greater ionization. A radio wave entering the Kennelly-Heaviside layer at an angle encounters regions of increasingly greater electron density, and consequently undergoes increasingly greater refraction or bending; this phenomenon is known as *progressive refraction*. It is quite possible for the bending to progress to the point where the radio wave travels horizontally with respect to the earth, then actually bends back toward the earth in the manner shown in Fig. 7B.

E, F_1 and F_2 Layers. Actually, the Kennelly-Heaviside layer is not as simple as is shown in Fig. 7B. The ac-



FIG. 7. These diagrams show how sky waves can be refracted (bent) back toward the earth by the Kennelly-Heaviside layer in the upper atmosphere.

tion of the sun upon the upper atmosphere is such as to produce three separate layers, one above the other. If we were able to measure electron densities, starting from the earth and going vertically upward, we would find that for the first thirty to forty-five miles there would be a negligible number of electrons. Continuing upward, the electron density would increase to the first maximum value at a height of about sixty miles, then gradually decrease. Higher up still, electron density would increase again and reach another maximum at about 125 miles. Farther up we would pass through another region \cup becomes quite brittle, and this appreciably affects the tone quality of the loudspeaker. The installation of a new cone or a new loudspeaker will usually provide a definite improvement in tone quality, with a corresponding increase in customer satisfaction at your work.

When replacing a loudspeaker with an exact duplicate replacement unit obtained from the receiver manufacturer or a radio supply house, all you need to be careful about is the connections. Make a rough sketch showing the terminals to which the voice coil and field coil leads of the old loudspeaker were connected, and use this as your guide for connecting the leads of the new loudspeaker.

The Output Transformer

The small radio part mounted on top of the loudspeaker in Fig. 11 is the *output transformer*. Its job, as you know, is to transfer audio signals from the audio output tube of the receiver to the loudspeaker in such a way that the loudspeaker gets the low-voltage, high-current signal it requires for most efficient operation.

Construction. The output transformer has two coils of wire, wound one over the other on an iron core. The primary coil contains many turns of fine wire. The secondary has fewer turns but uses heavier wire. The primary coil is connected to the output of the receiver, while the secondary coil is connected to the voice coil of the loudspeaker.

Open Coil. A break in either of the coils of wire in the output transformer will cause failure of a radio receiver. When such a break occurs, the serviceman says that the output transformer is *open*, and replaces the entire transformer. If the old transformer is held in place with rivets, he either drills out the rivets or cuts off

their spread-out ends with side-cutting pliers. There are usually only four connections to unsolder, so this is a fairly simple service job. Either rivets or nuts and bolts can be used for mounting the new transformer.

Once the trouble has been isolated to the audio output stage by professional servicing techniques, the output transformer can be checked by two simple ohmmeter measurements, one across each winding.

The Gang Tuning Condenser

In Fig. 12 is a clear illustration of another highly important large part in our superheterodyne receiver, the gang tuning condenser. It has two sections, which work with the two r.f. coils located underneath the chassis. A tuning condenser is never used by itself; it is always used with a coil, and tunes the coil to a particular frequency. You will find this interesting situation many, many times in radio different parts acting together to produce special results which the parts could not give separately.

One gang tuning condenser section is connected to an r.f. coil in the input circuit of the mixer-first detector input, and tunes the r.f. coil to the frequency of the desired incoming signal. The other gang tuning condenser section is connected to the r.f. oscillator coil in the oscillator circuit, and tunes this coil to a frequency exactly 456 kc. higher than the desired incoming carrier frequency.

Rotor and Stator. Each section of our gang tuning condenser in Fig. 12 has two important parts:

1. A rotor, which in this particular unit has eleven aluminum plates. Both rotors are mounted on the same condenser shaft.

2. A stator, which consists here of ten aluminum plates mounted on but insulated from the condenser frame. of gradually decreasing and then increasing electron density until we reached the third and final maximumdensity level at about two hundred miles up. Above this level the electron density decreases to a negligible value, for at this height in the stratosphere there is almost a total vacuum, and little ionization can occur. These three maximum-density layers are known as the E, F_1 and F_2 layers respectively.

The *E* layer is nearest the earth and is fairly constant in height; the F_2 layer is highest and may vary in height from 150 miles to 225 miles. This highest layer is most affected by the sun and hence it varies greatly in height and electron density from day to night, from summer to winter, and with con-



FIG. 8. Effect of the Kennelly-Heaviside layer upon the polarization of a sky wave.

ditions on the sun such as sun spots (solar eruptions). The F_2 layer also varies with latitude, and may be higher or lower around the equator than it is in the north temperate zone or at either pole. Oftentimes the F_1 (middle layer) and F_2 layers will merge to form what V is called an F layer. Less often, the E layer will separate into two layers, with one being known as the normal E layer and the other as the sporadic E layer.

A sky wave may be refracted back to earth by the lower portion of the Elayer, as shown by path 1 in Fig. 7C; on the other hand, it may be refracted by the upper part of the E layer as in path 2, may pass completely through the E layer and be refracted by the F_1 and F_2 layers as shown by paths 3 and 4, or may pass completely through all three layers as in path 5 and be lost in interplanetary space. As a rule, the longer wavelengths (broadcast band and longer wavelengths) are refracted by the E and F_1 layers, and shorter wavelengths are refracted by the F_1 and F_2 layers. The extremely short waves (ultra-high frequencies) pass through all three layers, and hence are not ordinarily received.

The angle θ in Fig. 7C (the angle between the sky wave and the earth) has a vital effect upon the path taken by the sky wave; the greater this angle, the more chance there is for the wave to get through the E layer and be refracted by the higher layers. A sky wave radiated at a high angle will be bent back to earth closer to the transmitting station than a wave radiated at a low angle and refracted by a lower layer. This statement is illustrated in Fig. 7D; radiation of the sky wave at the largest angle, along path 3, results in reception at point R_1 , which is near to station T. Radiation at a low angle, along path 1, gives reception at point R, which is a considerable distance from the station. Of course, the ground wave will be received in the immediate vicinity of the station in all cases.

Long-Distance Reception. The Kennelly-Heaviside layer makes possible long-distance reception of signals from low-power radio stations, for this layer bends radio waves back to the earth at points far beyond the range of reliable ground wave reception. These sky waves will be reflected skyward again by the earth, and will oftentimes be refracted by the Kennelly-Heaviside layer and reflected by the earth one or more additional times. Each path from earth to sky and back again is called a hop. Signals from station T in Fig. 7D may reach receiving point R in 1, 2 or 3 hops, as illustrated by paths 1, 2 and 3 respectively in Fig. 7D. As a rule, \checkmark signals on the longer wavelengths are sent with fewer hops than those on the shorter wavelengths. The best wavelength to use for communication between two given points depends considerably upon the conditions in the electron layers. The number of hops is controlled by the angle at which the sky wave leaves the station: the ideal number of hops at any time depends upon conditions in the upper layers. Where reliable transmission is essential, the same message is oftentimes broadcast simultaneously on several different wavelengths, each reflecting from a different layer and giving a different number of hops.

Peculiarities of Sky Wave Reception

Polarization of Sky Waves. If we analyze the electric field produced at receiving point R by the sky wave from station T in Fig. 8A, we find that it has doublet antenna at R in Fig. 8A, the antenna wire should of course coincide with the electric lines of force nearest point R in this diagram; amateur and commercial antenna systems are sometimes tilted to take advantage of this fact, but for home entertainment purposes it is impractical and unnecessary to provide means for tilting the antenna to a new angle each time a new distant station is tuned in.

Fading. For signals below 3 megacycles in frequency, the sky wave is refracted and bent back to the ground almost immediately after its entrance into the lowest electron layer (the Elayer). If there is an appreciable amount of high-angle radiation (sky waves) from the station, receiving an-



FIG. 9. This diagram shows why fading sometimes occurs within 50 miles of a broadcast station.

a vertical component $E_{\mathbf{v}}$ and a horizontal component $E_{\rm H}$, as shown in Fig. 8B. When θ (the angle of radiation) is more than 45°, the $E_{\rm H}$ component will be the stronger, and a horizontal receiving antenna will give better signal pick-up than a vertical antenna. When a radio wave travels a long distance with a single refraction from the Kennelly-Heaviside layer, θ may be less than 45°: the vertical component of the electric field will then predominate, and a vertical antenna will give better signal pick-up provided there is no man-made interference near the receiver. When interference is present, a horizontal antenna is invariably more satisfactory, for the horizontal component $E_{\rm H}$ of a sky wave is usually strong enough to give a high signalto-noise ratio.

For maximum signal pick-up with a

FIG. 10. Long-distance communication via short-wave radio depends en-tirely upon sky waves. The transmitter engineer chooses operating condi-tions which will make the sky wave come back to earth at the desired reception area.

tennas about fifty miles away from the station will pick up the ground wave directly and the sky wave after refraction, as indicated in Fig. 9. The sky wave, traveling a considerably longer distance, may arrive in phase with the ground wave, in which case there is reinforcement and maximum signal pickup; more likely, the two waves will be partially or completely out of phase, giving partial or complete cancellation of the signal. The phase relationship may vary from instant to instant, causing fading, for the Kennelly-Heaviside layer sometimes shifts up and down slightly. Briefly, then, sky waves cause fading at receiving locations near the limits of reliable ground wave reception from a given station because the phase relationship between the sky and ground waves changes continually due to refraction of the sky wave by the rapidly - shifting Kennelly - Heaviside Tayer, giving partial or complete cancellation of the ground wave.

If the shifts which cause fading occur in less than one-tenth of a second, the a.v.c. system in the average receiver will not be able to compensate for them and the variations in signal strength will be annoyingly noticeable at the receiver loudspeaker. Even when fading takes place at a slow rate, the automatic changes in receiver gain may result in noticeable alternate periods of clear and noisy reception.

Fading is generally less severe during the daytime. The Kennelly-Heaviside layer is then so dense that radio waves at broadcast band frequencies are almost completely absorbed as they enter the lower regions of the layer. Fading at distances of from 40 to 60 miles away from a broadcast station can be kept at a minimum by designing the transmitting antenna to radiate most of its energy along the ground; vertical antennas which are about $\frac{5}{8}$ wavelength high are best suited for this purpose.

Skip Distance. At frequencies of from 3 to 30 megacycles, the sky wave is used almost exclusively for communication purposes. The transmitting antenna is in this case designed to radiate most of its signal into the sky, and may have a vertical radiation pattern like that in Fig. 10A. Notice that maximum signal intensity occurs at a definite angle of radiation with the ground, and that radiation is concentrated over a definite angular distance which is known as the beam angle, There may be enough radiation along the ground in a case like this to provide adequate reception for distances of perhaps ten miles away from the station, but most of the energy is radiated into the sky, where it is either bent back to the earth or lost through absorption. This type of radiation pattern for a transmitting antenna provides re-

ception over a more or less restricted area at a great distance away from the station; the distance and the area both depend upon the width of the beam, the angle of radiation, the frequency of transmission and upon conditions in the Kennelly-Heaviside layer. The distance between the maximum limit of ground wave reception and the closest point at which the sky wave returns to earth is commonly known as the skip distance (see Fig. 10B). Different frequencies must be used for each distance covered, and frequency must be changed at night to compensate for the shift in the height of the electron layers and the resulting change in the skip distance, as each frequency is acted upon differently by the layers. As a general rule, high transmitting frequencies are best for daylight and lower frequencies are better for night transmission. Furthermore, radio transmission is better in winter than in summer, for in winter there is less absorption of signals by the electron layers.

Selective Attenuation. We must not overlook the fact that the transmission of voice and music is accomplished by sending a number of side-band frequencies along with an r.f. carrier. Each side frequency is acted upon by the Kennelly-Heaviside layer in a slightly different manner, and consequently the various frequencies may take various paths back to the earth or may be attenuated differing amounts. The result is that side frequencies in the received signal may have a different strength relationship to the carrier frequency than they had at the transmitting station, causing amplitude distortion in the demodulated audio signal; this cannot be corrected in the receiver. Since conditions in the Kennelly-Heaviside layer fluctuate quite rapidly, reception may be distorted severely at one moment and may become clear in the next instant, particu- \checkmark larly when the receiver is located near

the boundary between a skip area and a reception area. \checkmark

Receiving Antenna Problems

Even though a transmitting station is designed to "lay down" a strong signal in a definite receiving area, part or all of this energy will be wasted unless the receiving antenna is properly designed. Three important factors are involved in the selection and installation of a receiving antenna:

1. That portion of the antenna which intercepts the electric and magnetic components of the radio wave must have maximum possible signal pick-up. This involves consideration of the directional properties of the antenna, the polarization of radio waves from favorite stations, and the frequency range over which signals are to be picked up.

2. The signal energy picked up by the antenna must be transferred to the receiver with a minimum of loss over the desired range of carrier frequencies. This involves matching the transmission line impedance to the impedance of the antenna and to the impedance of the receiver input, to securé maximum transfer of energy from the amtenna to the receiver.

3. There should be a maximum signal-tonoise ratio at the receiver input. This is usually secured by designing the antenna system so it will accept a minimum of noise signals.

Directivity of Receiving Antennas

The directional characteristics of a receiving antenna system depend upon a number of important factors: 1. The current and voltage distribution along the antenna; 2. The position of the antenna with respect to ground; 3. The position of the pick-up section with respect to other elements of the antenna, if the system is made up of more than one wire; 4. The electrical characteristics of the ground near the antenna.*/

Current and Voltage Distribution of Half-Wave Doublet Antennas. A single straight half-wave doublet antenna like that in Fig. 11A, located in free space (away from the influence of the earth and other objects) and fed with energy either by a direct connection to a transmitter or by radiation of power through space from a distant transmitter, acts like a number of small elements of inductance and capacity connected together in the manner shown in Fig. 11B. If a.c. power is fed into the center in the case of a half-wave doublet transmitting antenna, the cur-



FIG. 11. Characteristics of a simple half-wave transmitting or receiving antenna in free space.

rent encounters small elements of inductance which serve to choke it and reduce its value, and for each inductance there is an elemental capacitive path which draws off a portion of the antenna current and brings it to a corresponding point on the other half-section of the antenna. The result is that the r.f. signal current is the greatest at the center of a half-wave receiving antenna, and gradually drops to zero as it approaches the extreme ends. Across cach elemental inductance there will be a voltage drop due to current flow through it, and those elements nearest the center will naturally have the highest voltage drops. The result is that the r.f. voltage with respect to the center of the antenna increases rapidly as we move outward from the center, then increases more gradually to a maximum value at the ends; furthermore, ∇

^{*}Items 3 and 4 are not covered in this lesson since they are ordinarily neglected when designing and installing antennas for home radio receivers. These factors are extremely important in connection with amateur and commercial receiving antennas and with television receiving antennas, and are taken up in advanced lessons dealing with these subjects.

the ends will be opposite in polarity at any instant and will reverse in polarity once for each cycle. The curves in Fig. 11C show this current and voltage distribution; they tell, for example, that the effective (r.m.s.) value of the r.f. current at point P is proportional to the distance NP, and the r.m.s. value of voltage at point P is proportional to the distance MP. Notice that both curves are sinusoidal (having a sine wave shape), with maximum current at the center and maximum voltage at the ends. (Since the center of the antenna has zero voltage, it may be considered at ground potential and all voltages measured with respect to the center rather than to ground.)

This half-sine-wave distribution of current and voltage will be obtained for any doublet transmitting or receiv-



FIG. 12. Characteristics of a grounded vertical quarter-wave antenna. A perfectly conductive ground is assumed.

ing antenna in free space whenever the length of the antenna wire is equal to one-half the wavelength of the signal. This condition can occur at only one signal frequency; at that frequency the antenna radiates or picks up a maximum signal, and is said to be at resonance.

The essential requirement of a resonant receiving antenna is that it act as a pure resistive source. This requirement can be met by making the antenna one-half wavelength long or some multiple of a half wavelength (2, 3, 4, 5, etc., times as long). An antenna which is not a multiple of one-half wavelength for a particular radio wave will have either an inductive reactance or a capacitive reactance which raises its impedance and reduces the amount of current which can be delivered to the receiver. It is possible to introduce either capacity or inductance in the antenna circuit in order to cancel out the existing reactance; this *tunes* the antenna to resonance, making it purely resistive and giving at its ends the zerocurrent, maximum-voltage condition which results in maximum signal pickup.

Current and Voltage Distribution of Grounded Vertical Antennas. If a vertical antenna is grounded at one end, it will be at resonance when only 1/4 wavelength long, for the earth serves to duplicate the effect of the other 1/4 wavelength. A quarter-wave grounded vertical antenna is shown in Fig. 12A; a receiving antenna at point $P_{\rm s}$ in space will receive a radio wave directly from point P_A on the transmitting antenna, and will receive another radio wave directly from point P_{A} by reflection from the ground. This reflected wave appears to come from a point P_{I} which is located the same distance below the surface of the earth as P_{A} is above the earth. In the language of radio engineers, we have an image antenna in the ground; although it is a purely imaginary structure, the observed effects are exactly the same as if it were present.

The current and voltage distribution curves for a grounded quarter-wave vertical antenna are shown in Fig. 12B; note that current is zero and voltage a maximum at the upper end of the antenna, just as with a half-wave antenna. (Remember that these curves indicate r.m.s. values; the current and voltage at each point along the antenna are varying in a sine wave manner at the signal frequency.) A grounded vertical antenna is at resonance whenever its height is some odd multiple of 1/4 wavelength; it can therefore be $\lambda/4$ (λ is the symbol for one wavelength), $3/4\lambda$, $5/4\lambda$, etc. An untuned vertical antenna (an antenna

whose height at a particular signal frequency is not a multiple of $\lambda/4$) can be tuned to resonance, making it purely resistive and giving maximum signal pick-up at that signal frequency, by adding either a coil or a condenser in series.

Typical resonant receiving antennas are shown in Fig. 13; current distribution curves for these are shown in dotted lines. At A, B and C are grounded $\lambda/4$ antennas, while at D and E are ungrounded $\lambda/2$ antennas; in each case the free ends have zero current. A receiving antenna which is at resonance when straight will remain at resonance even though a part of it is bent at an angle. A resonant half-wave antenna for reception at 49 meters should be $49 \div 2$, or 24.5 meters (81) feet) long. A grounded quarter-wave antenna which is to be resonant at this same wavelength should be about 40.5 feet long.

A quarter-wave antenna which is resonant at 98 meters will be a halfwave antenna for signals at 49 meters and will be a full-wave antenna for signals at 24.5 meters. At wavelengths other than these values we can secure signal pick-up approaching that of a resonant antenna by introducing into the antenna system a condenser which will tune it to a lower wavelength, or by inserting a coil which will tune it to a longer wavelength. This practice is used quite often, especially when the antenna does not have the correct length for resonance at a desired frequency. For example, if an 81-foot length of wire is needed for a center-fed half-wave antenna and only 61 feet of room is available, the surplus 20 feet of wire can be wound in two coils of equal size, and one placed in series with each section of the antenna, near the center.

Directional Characteristics. Although the shape of a receiving antenna has little effect upon resonant conditions, shape does affect the directional characteristics of an antenna. If an antenna is short in comparison to the wavelength of the signals which it picks up (most of the antennas used for 500 to 1,500 kc. broadcast band reception are in this class), best signal pickup is secured when the electric lines of force associated with the radio wave arrive parallel to the antenna wire. Thus, the vertical portion of a short antenna will respond best to vertically polarized electric fields, and the horizontal portion of a short antenna will receive horizontally polarized waves best. Only short antennas (shorter than one-quarter wavelength of the radio wave which is intercepted) have these properties, for they are acted



FIG. 13. These curves show that current distribution I along a quarter-wave or half-wave antenna is essentially independent of the shape of the antenna.

upon directly by the electric field and are not greatly influenced by surrounding objects or by the ground.

Radiation Patterns for Receiving Antennas. When the length of an antenna is one-fourth the wavelength of the signal picked up (or is longer than 1/4 wavelength), special patterns must be used to show the directional characteristics. These patterns are called radiation patterns because they are usually secured by making measurements while the antenna in question is connected to a transmitter and is radiating signals, but each pattern applies to a particular antenna regardless of whether it is used for transmitting or receiving purposes. A horizontal radiation pattern for a receiving antenna indicates the effectiveness of signal pickup in any direction along the ground. A vertical radiation pattern for a re-
ceiving antenna indicates the effectiveness of signal pick-up at various angles to the ground in one vertical plane passing through the center of the antenna system.

In Fig. 14A is shown the radiation pattern in a vertical plane for a

tern indicates that a vertical antenna receives ground waves equally weh from all directions along the surface of the earth. When a grounded vertical antenna is shorter than $\lambda/4$, the center of a horizontal wire can be connected to the top of the vertical antenna to



FIG. 14. Current distribution curves (dotted lines) and vertical radiation patterns (thin solid lines) at one particular frequency for grounded vertical antennas having various physical lengths.

grounded vertical $\lambda/4$ antenna; this pattern will be the same for vertical planes in all directions (north, east, south or west) for this particular antenna and height. As antenna height is increased above $\lambda/4$ the radiation pattern flattens out, indicating improved pick-up of the ground wave, as shown at *B*, *C*, *D* and *E* in Fig. 14. When antenna height becomes more than $\lambda/2$ (.5 λ), a lobe or ear directed up into the sky begins to form, however; this indicates that the antenna is beginning to favor reception of sky waves. If the height of the antenna is form the T antenna shown in Fig. 15B; this does not affect the horizontal radiation pattern, but it does serve as a load which helps to tune the antenna to resonance. The current distribution curves show that currents are zero at the ends of the horizontal portion and increase gradually up to point a, where the two currents combine to flow down the vertical portion of the antenna These currents are equal for a symmetrical T antenna, and since they flow in opposite directions toward point ain Fig. 15B they do not alter the nondirectional characteristics of the verti-



Fig. 15. Horizontal radiation patterns for grounded vertical, T and inverted L types of antennas.

increased beyond .75 λ , only sky waves will be received effectively. Remember that these patterns are for grounded vertical antennas; when a vertical antenna is elevated above the ground, the patterns become quite different.

The horizontal radiation pattern for a quarter-wave grounded vertical antenna is shown in Fig. 15A. This patcal portion. The T-type antenna is particularly desirable when it is impractical to erect a sufficiently high vertical antenna to give resonance. Resonant conditions produced by the antenna itself always give better signal pick-up than resonance produced by inserting coils or condensers.

A vertical antenna shorter than $\lambda/4$

can be increased in effective length by connecting to its top a single horizontal wire as shown in Fig. 15C, giving an inverted L antenna. The horizontal portion serves to load the vertical portion; if its length is properly chosen, it will tune the antenna system to resonance at one desired signal frequency, giving maximum possible current at



FIG. 16. Radiation patterns for $\lambda/2$ and λ antennas in free space. The patterns are the same for all planes passing lengthwise through the antennas.

the receiver. Radio waves arriving from the direction in which the horizontal section is pointing are favored, giving a slightly directional radiation pattern in the horizontal plane, as shown in Fig. 15C. The natural wavelength of an L or T type antenna is determined by measuring from the ground to the end of the longest horizontal section.

The radiation pattern of a theoretically perfect half-wave antenna in free space will be the same in any plane that it receives best from its sides, and has practically no signal pick-up from its ends. Practical doublet antennas have somewhat similar directional characteristics, and consequently should be kept broadside to a station from which maximum signal pickup is desired.

The radiation pattern for a fullwave (λ) antenna in free space, shown in Fig. 16B, resembles somewhat a four-leaf clover, indicating that best reception is restricted to four narrow beams, and that there is no reception either from the ends (in the plane of the antenna) or from the sides (at right angles to the antenna). More and more of these lobes appear in the radiation pattern as antenna length is increased in multiples of $\lambda/2$, with the lobes becoming longer and narrower; antennas longer than $\lambda/2$ are therefore unsuited for general reception in all directions.

It is practically impossible to erect a horizontal doublet receiving antenna which is high enough above the ground so it will act as if it were in free space. The average radio receiver owner would have difficulty in erecting a horizontal antenna which is higher than about 50 feet above the ground (about 15 meters). This would make the antenna $\lambda/4$ above the ground for a 60-





passing through the antenna wire and will be a figure-of-eight pattern as shown in Fig. 16A. A three-dimensional radiation pattern, showing the pick-up characteristics in all directions, would appear like a round "doughnut," with the antenna wire running through the doughnut hole. The outstanding fact about a $\lambda/2$ antenna in free space is meter wave. The radiation patterns in two vertical planes at right angles to each other for a $\lambda/2$ horizontal doublet antenna mounted a distance of $\lambda/4$ above the ground are shown in Fig. 17A. These patterns tell us that sky wave reception is good and is essentially the same in all directions. The patterns might lead you to expect poor

ground wave reception, but ground waves are usually tilted sufficiently to induce ample voltage in an antenna of this type, giving quite good ground wave reception also. When the halfwave doublet antenna in Fig. 17A is raised another 1/4 wavelength, making it $\frac{1}{2}$ wavelength above the ground, a decidedly directional radiation characteristic is obtained, as indicated in Fig. 17B. Sky waves coming in at a reasonable angle (the usual condition for these waves) are picked up quite well, but waves arriving at high or low angles of elevation are poorly received. An antenna such as this would not be good for reception of signals broadcast by an airplane flying overhead, nor for reception of ground wave signals unless they were well tilted.

The angle at which maximum energy reaches a receiving antenna is the same as the angle at which maximum energy leaves the transmitting antenna. In other words, if a given transmitting antenna radiates maximum energy into the sky at a definite angle of elevation, maximum pick-up of signals from that antenna will be obtained with a receiving antenna which has maximum pickup at that same angle of elevation.

Conclusion. Keep the following facts in mind when installing receiving antennas:

1. A simple antenna having a pick-up portion which is $\lambda/4$ meters long or less and in one straight line gives best reception of radio waves whose electrical lines of force are parallel to the pick-up wire.

2. Horizontal doublet antennas having pick-up sections with a total length of $\lambda/2$ or less give best pick-up of signals arriving broadside to the antenna, but give fairly good pick-up from other directions as well at normal antenna heights above ground.

Coupling The Antenna To The Receiver

Now that we have seen how the pick-up portion of an antenna can be made to produce maximum voltage and

maximum current for a given radio wave, we are ready to consider how maximum signal can be transferred from the antenna to the radio receiver. For antennas less than a quarter-wave long, the solution is quite simple, involving merely a direct connection as shown in Fig. 18. The entire length of the antenna system will pick up signals, and consequently length b-c must be added to length a-b when determining the operating wavelength of the system. For a $\lambda/4$ system, the current will be a maximum at the ground end; if the receiver is connected close to ground, maximum current will flow through primary coil P and will induce



the greatest possible voltage in secondary coil S. A long ground wire between receiver and ground moves the primary coil to a point where current is less. reducing the efficiency of the antenna; furthermore. an excessively long ground wire picks up noise and thus lowers the signal-to-noise ratio. Keep the ground wire as short as possible, and increase the length of the antenna itself up to 1/4 wavelength if maximum pick-up is required at a given wavelength.

A $\lambda/4$ antenna for best reception of a 1,500 kc. (200-meter) signal should be 50 meters (165 feet) long. This length is rarely attained in the average receiving antenna, but a coil can be inserted in the lead-in wire (at point x in Fig. 18) to offset this lack of length. On short-wave bands, however. even a 50-foot antenna is too long for $\lambda/4$ operation. When an antenna is only slightly longer than $\lambda/4$ (not more than about 3/8 wavelength long), the excessive length can be offset by inserting a variable condenser at point xand adjusting for maximum signal pick-up from the desired station (this tunes the antenna to resonance, making it equivalent to a $\lambda/4$ antenna at that station frequency). Tuning of an antenna complicates the operation of a receiver, so it is customary to design the primary coil so it will make the average broadcast antenna tune approximately to the broadcast band wavelengths, and provide means for reducing the size of this coil on short wave bands.

Voltage Feed For a Doublet Antenna. With a doublet antenna, it is quite possible to connect two wires (a transmission line) to points which are equally spaced on each side of the center, as shown in Fig. 19A, in order to transfer to the receiver a portion of the r.f. voltage which is developed by the antenna. The voltages at P_1 and P_2 are opposite in polarity at any instant of time (see Fig. 11C), and hence the voltage between P_1 and C will always add to that between P_2 and C. This connection is commonly referred to as a voltage feed to the receiver.

Current Feed. When a doublet antenna is cut at its center and two wires are used to connect the cut ends to the receiver, we have what is known as current feed to the receiver. Antenna current flows along one half of the pick-up section, down one lead-in wire to the receiver, through the antenna coil of the receiver, up through the other lead-in wire, and finally out over the other half of the pick-up section. Since current in a $\lambda/2$ or shorter antenna is a maximum at the center, this connection gives maximum current through the antenna coil of the receiver. Current feed is widely used with

antennas for all types of receivers.

A $\lambda/2$ antenna which is designed for 80 meters will be a full-wave antenna at 40 meters, a $3/2\lambda$ antenna at 30 meters, and a 2λ antenna at 20 meters. Let us see if an antenna like this will work effectively at all three wavelengths with current feed. Figure 20 shows the current distribution for these four operating conditions; these curves show that current is a maximum at the center only for operation as a $\lambda/2$ or $3/2\lambda$ antenna. Current at the center is zero for λ or 2λ operation, which means that a current feed at the center will theoretically give no signal transfer to the receiver under these conditions; it



FIG. 20. Current distribution curves for $\lambda/2$, λ . 3/2 λ , and 2 λ operation of a doublet antenna.

would be necessary to move the tap to an off-center point of maximum current in order to secure efficient signal pick-up. This explains why a single $\lambda/2$ antenna which is used for all-wave reception and has a current feed connection to the receiver will give excellent results at certain frequencies and will be very inefficient at other frequencies.

Transmission Lines

Antenna Resistance and Impedance. Any antenna which is tuned to resonance will act as a resistance source of r.f. signals; at the center of a $\lambda/2$ antenna this a.c. resistance will be about 72 ohms. For a grounded $\lambda/4$ antenna, the a.c. resistance at the ground (measured between the antenna lead-in wire and the ground wire) will be about 36 ohms. A voltage-fed $\lambda/2$ antenna will have a resistance considerably higher than 72 ohms; if the transmission line connections are made near the ends of

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FIG. 12. View of the gang tuning condenser in our superheterodyne receiver, with important parts identified.

The rotor of each tuning condenser section *rotates*, but the stator of each section is stationary, staying in the same position all the time.

How the Capacity Varies. When the shaft of the gang tuning condenser is turned in one direction, the rotor plates go farther in between the stator plates, without ever touching the stator plates. This increases the electrical size or *electrical capacity* of each tuning condenser section.

Turning the shaft in the opposite direction brings the rotor plates out from the stator plates, reducing the electrical capacity of each section.

It is this chan e in the electrical capacity of each gang tuning condenser section which tunes the receiver to different stations. In later lessons you will learn exectly how this is done.

Dial Cord. The tuning control knob in our superheterodyne receiver is quite some distance from the can tuning condenser shaft, as you have probably noticed already in Fig. 12. Now, what makes the tuning condenser shaft rotate when we turn the tuning control knob? Also, what makes the dial pointer move to the correct dial number for the station whose program is coming from the loudspeaker?

Figure 13 shows the answer—simply a length of braided dial cord and a few pulleys. The cord runs around a small pulley on the tuning control knob shaft, around a large pulley on the gang tuning condenser shaft, and around two idler pulleys. A spring inside the large pulley keeps the cord tight.

The pointer is attached to the dial cord at the position shown. When you turn the tuning control knobin a clockwise direction, the sliding pointer moves to the right along the tuning dial, and the gang tuning condenser shaft turns in a clockwise direction.

The average radio set owner never sees this cord-and-pulley system, because it is at the front of the chassis, and cannot be seen when the chassis is in its cabinet. For this reason, a the antenna, this resistance may be as high as 5,000 ohms. At off-resonant conditions, reactive components are added to these resistance values, making the impedance of an untuned antenna quite high.

Surge Impedance of Transmission Lines. Any two-wire transmission line is a combination of elemental resistances, inductances and capacities arranged as shown in Fig. 21A. At low audio frequencies the inductances and capacities have a negligible reactive effect and the line simply has a d.c. resistance distributed along its length due to the resistance of the wire itself. At radio frequencies, however, the elemental inductances and capacities have much more effect upon the characteristics of the line than do the elemental resistances, and we can for practical purposes neglect entirely the presence of these resistances. The values of these elemental inductances and capacities control one extremely important characteristic rating of a transmission line—its surge impedance usually designated as Z_0^* . The inductance values are determined by the diameter of the transmission line wires and the spacing between them; the capacity values are determined by the diameter of the wires, the spacing between the wires, and by the nature of the dielectric materials between the two wires. Neither the frequency of the radio signal nor the length of the line ordinarily has any effect whatsoever upon the inductance and capacity values. This means that the surge impedance of a transmission line depends only upon the construction of a unit length of a transmission line, and is

*This impedance is also known as the *characteristic impedance* of a transmission line; "characteristic impedance" is a more readily understood term, but "surge impedance" is the term more widely used. The value of this impedance depends upon the ratio of elemental inductance to elemental capacitance. the same for any length of line and for any frequency. Increasing the distance between the two line wires will increase the surge impedance; reducing the diameter of the wires will increase the surge impedance; the use of a dielectric material which increases the capacity between the two wires will reduce the surge impedance.

If we were to measure the impedance between the input terminals of an extremely long (infinitely long) transmission line, we would get a value which is equal to the surge impedance of that line. Furthermore, this impedance would be purely resistive, for the large numbers of elemental inductances and capacities distributed over this infinitely long line make it act as a broadly tuned resonant circuit having the same impedance at all frequencies. This fact is illustrated in Fig. 21B.

If we were to measure the input impedance of a practical transmission line (seldom more than one or two wavelengths long in radio work), we would probably get a value considerably different from the surge impedance of the line; actually the value of impedance which we measure would depend upon the length of the line and upon how it was terminated (the nature of the load at the output end of the line). It is possible to compute the input impedance of a transmission line having any length and any type of load at its output end, but the mathematical procedure involved is highly complicated and for all practical receiving antenna purposes is quite unnecessary.

We are not concerned with this input impedance of a practical transmission line, because there is a simple way to make this input impedance equal to the surge impedance of the transmission line. If the output end of a transmission line is terminated in a load having an impedance exactly equal to the surge impedance of the transmission line, the input impedance of that transmission line will then be equal to the surge impedance of that line regardless of its length; this is illustrated in Fig. 21C. If the input end of this line is now connected to a signal source having an impedance equal to the surge impedance of the transmission line, the impedance at each end of the transmission line will be equal to the surge impedance Z_0 regardless of the length of the line. This phenomenon is widely used in the design of transmission lines for connecting receiving antennas to receivers, for it simplifies construction considerably and gives at



FIG. 21. Important facts concerning the behavior of transmission lines.

each end of the transmission line an impedance match which results in maximum transfer of power from the antenna to the receiver.

Typical Surge Impedance Values. A transmission line made up of two No. 18 wires spaced 2 inches apart will have a surge impedance of about 560 ohms. Two No. 10 wires separated by 2 inches have a surge impedance of about 440 ohms. No. 18 and No. 10 wires separated by 3 inches will have surge impedances of 610 ohms and 480 ohms respectively. These so - called "open" transmission lines are rarely used with receiving antennas, but are widely used with transmitter installations. When two No. 18 rubber-covered wires are twisted together or when one rubber-covered wire is placed inside a braided metallic covering, the surge impedance Z_0 will range between 70 and 150 ohms: by careful design, this

impedance can be made equal to 72 ohms, making it possible to couple a current-fed $\lambda/2$ antenna directly to the transmission line, securing a perfect impedance match and maximum transfer of power from the antenna to the transmission line. If the receiver is made to serve as a 72-ohm load on the other end of this line, we will have maximum transfer of power from the antenna to the receiver regardless of the length of the transmission line.

It is not essential that a 72-ohm transmission line be used for maximum transfer of power. An r.f. impedancematching transformer, usually called an antenna transformer, can be used at the antenna to match the antenna impedance to the surge impedance of the transmission line regardless of what their respective values may be, thereby securing maximum transfer of energy. Another impedance - matching transformer can be used at the receiver end to match the receiver input impedance to the transmission line surge impedance. When a receiver is designed for use with a definite antenna system, the first r.f. transformer in the receiver is designed to match the surge impedance of the transmission line, eliminating the need for a separate matching transformer.

Noise Reduction

It is useless to erect an antenna which will deliver a strong carrier signal to a receiver if this signal is "blanketed" by local man-made interference noise which is also picked up by the antenna. As a general rule, manmade interference produces electric fields which are *perpendicular* to the ground, just like the fields produced by a vertical transmitting antenna. If a horizontal antenna with a vertical down lead is used for reception, most of the interference noise will be picked up by the down lead; very little will be picked up by the horizontal section. The chief problem in securing a noisereducing antenna is that of preventing the down lead or transmission line from picking up noise signals.

With a doublet antenna, the transmission line inherently rejects noise signals. The two down leads of the transmission line are normally twisted together in the manner shown in Fig. 22A, so that very little electric field can exist between adjacent wires in the transmission line. Any noise voltages which are induced in these vertical wires cause currents to flow in the same direction in both wires at any instant of time. Noise currents therefore flow in opposite directions through the primary of the receiver transformer to the center tap and then to ground; as long as these two noise currents are balanced, they cannot produce any flux in the receiver transformer. Some noise signals may be picked up by the entire antenna system acting as a vertical mast: an electrostatic shield is usually placed between the primary and secondary windings of the receiver input transformer to prevent these noise signals from being transferred to the secondary winding.

The antenna and receiver transformers in Fig. 22A can give a perfect impedance match at only one frequency; the tuning action can, however, be made sufficiently broad by means of tight coupling in the transformers to give satisfactory reception over a wide range of frequencies.

Noise-reducing antenna systems for the 500 kc. to 1,500 kc. broadcast band are invariably of the single-wire horizontal type; a typical noise-reducing broadcast antenna is shown in Fig. 22B. Since limitations of space in the vicinity of the average home would normally prevent the use of a horizontal pick-up section which is $\lambda/4$ meters in length, a more reasonable length of from 50 to 100 feet is usually employed, and the primary of the an-

tenna transformer is designed to bring this length up to $\lambda/4$ for the highest frequency which the antenna is to pick up. The transmission line in this case is a single insulated wire surrounded by a flexible metal loom or shield; this shield is connected to the ground terminal of the receiver. A twisted 2-wire transmission line may be used if proper ground connections are made at the receiver, and the transmission line is properly connected to the antenna. transformer. For maximum reduction of noise and the most favorable distribution of current in the antenna, the shielding metal loom must be grounded near the antenna, either to a vent pipe,



FIG. 22. Two types of noise-reducing antennas.

to the grounded metal roof of the building, or to a grounded gutter. Additional grounding of the shield at several points is optional, but should be done whenever the transmission line runs conveniently near a grounded object. This type of noise-reducing antenna may also be designed for allwave reception.

Even though the vertical lead-in wires of an antenna system are carefully shielded against man-made noise fields, there is still a possibility that noise will be picked up by the antenna system. This is due to the fact that the electric fields produced by man-made interference may not always be exactly perpendicular to the earth; if they are slanted any appreciable amount and extend an appreciable distance above the ground, they will induce noise voltages in the horizontal pick-up wires. When a noise-reducing antenna system fails to give a satisfactory signalto-noise ratio, the usual procedure is to change the position of the horizontal pick-up portion of the antenna, such as by increasing its height, changing its direction, or moving it to a location farther away from the source of noise.

Noise-Reducing Loop Antenna. Another solution to the problem of noise fact, E may be as much as 70 times stronger than H. Noise interference sources are generally so close to the receiver that the vertically - polarized electric component E of the noise is much stronger than the magnetic component of noise, whereas the E and Hcomponents of the desired radio signals are equal. If we can remove the electric component E of both the noise signals and the radio signals, depending upon magnetic components only for reception, we will then have eliminated the



FIG. 23 (Left). Typical shielded loop antenna (General Electric Beamascope) designed for installation inside a console model home radio receiver. Part of the Faraday shield is cut away to show the loop winding and its wood frame inside. This unit is rotated to a position of minimum noise pick-up at the time the receiver is installed in a home.

Courtesy Zenith Radio Corp.

Above: A compact shielded loop antenna mounted at the rear of this 5-tube superheterodyne table model receiver (Zenith Wavemagnet) eliminates the need for antenna and ground connections. The flat loop is wound on a spider-web type coil form, and is mounted between two flat shields woven with vertical copper wires and horizontal insulating cords. All shield wires are connected together at the bottom and grounded to the receiver chassis.

interference is the use of a shielded loop antenna which is built right into the cabinet of a radio receiver. To understand how this shielded loop can reduce noise signals, we must first consider the relative strengths of the electric and magnetic fields associated with radio waves. At points outside of the induction field produced by a transmitting antenna or source of noise interference, the electric and magnetic fields associated with the radiation field (the radio wave) are essentially equal. Close to the source, however, the electric component E is much greater than the magnetic component H; in noise almost completely without impairing reception of the desired radio signal. The chief purpose of the grounded Faraday shield which surrounds some types of built-in loop antennas is therefore to prevent electric components of both station and noise signals from affecting the receiver.

One commercial version of this shielded loop arrangement is shown in Fig. 23. The Faraday shield is in the form of a closed cylinder with sheet metal discs covering the top and bottom faces and with the sides covered with a coarse woven material in which the vertical threads are wire and the



horizontal threads are non-conducting fiber. Every vertical wire makes contact with the top metal disc, but only one of these vertical wires makes electrical contact with the bottom disc: this construction eliminates closed circuits in the vicinity of the loop, giving maximum pick-up by the loop of the magnetic component of radio signals. The shield is grounded, and consequently all signal and noise currents induced in its vertical wires by the Ecomponents are led off to ground without affecting the receiver. The wellknown directional characteristics of a loop are utilized to give additional noise rejection; the loop is rotated for minimum noise, so its line of minimum pick-up is in the direction of the noise source. No external antenna or ground connections are necessary when one of these shielded loops is installed, but an outdoor antenna can be connected to the loop or in place of it if distant reception is desired. A loop of this type can be built right into the cabinet of a radio receiver, and can be tuned over the entire broadcast band by a main tuning condenser section which is adjusted to track with the oscillator; this tuning to resonance for each station increases the sensitivity of the loop very greatly.

All-Wave Antennas

Either horizontal doublet antennas or single-wire horizontal antennas with end feed may be used for all-wave reception. The doublet is ordinarily preferred, for its balanced two-wire transmission line gives inherent rejection of vertically-polarized noise signals. Both types of all-wave antennas favor slightly the reception of signals from certain directions, but these directional characteristics can ordinarily be neglected when installing an all-wave antenna for the average radio listener. It is far more important to place the horizontal pick-up section of the antenna at right angles to a nearby power line, in order to secure rejection of horizontally-polarized power line noise signals.

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Types of All-Wave Antennas Commonly Used. Three distinct types of antenna systems are in common use for all-wave reception over the entire range from about 18 meters (16.5 megacycles) to 545 meters (550 kc.). These are:

1. An antenna system made up of two or more doublet antennas, each tuned to a different wave-length. With the customary broadness of tuning in doublet antennas, fairly uniform pick-up is obtained over the desired range. A typical system might employ three half-wave doublets, tuned respectively to 16, 25 and 49 meters.

2. A system employing a single horizontal doublet antenna cut at its exact center for current feed, and with a length which will keep it less than a full-wave antenna at any frequency in the range to be received. For example, if the receiver is to go down to 18 meters, a 50-foot horizontal doublet can be used, for it will be a full-wave antenna at about 15 meters and will have sufficient current at its center at 18 meters. At 30 meters this antenna would become a $\lambda/2$ unit; at all longer wavelengths it will be shorter than $\lambda/2$, and will have maximum current at its center even though the current is gradually reduced due to off-resonant conditions as wavelength is increased.

3. A system employing a single horizontal wire having a properly grounded antenna transformer at one end for current feed, and with a length which will keep the effective pick-up portion of the system less than a half-wave long at any frequency in the range to be received. For example, if the shortest wavelength at which reception is desired is 18 meters, a 25-foot horizontal wire would very likely be used (this would be a halfwave antenna at 15 meters).

An all-wave doublet antenna gives maximum pick-up at a frequency which makes it $\lambda/2$. Pick-up decreases gradually above $\lambda/2$, for the current at the center of the antenna goes down. At λ operation the current at the center is zero, and there is practically no signal pick-up. Above λ , pick-up increases again to a maximum at $3/2\lambda$, then drops to zero again at 2λ . All this means that a doublet antenna will pick up stations having frequencies higher than that for λ operation, but reception will be erratic due to the highly directional and continually changing radiation patterns and to the dead spots in reception which occur at λ , 2λ , 3λ , 4λ , etc., operation.

A single-wire antenna acts in much the same way above its efficient operating length of $\lambda/4$; there will be dead spots in reception at $\lambda/2$, λ , $3/2\lambda$, 2λ , $5/2\lambda$, etc., for these operating lengths give zero current at the antenna transformer. Pick-up will be good at $\lambda/4$, $3/4\lambda$, $5/4\lambda$, $7/4\lambda$, etc., for these operating lengths give high currents at the antenna transformer.

Multiple Doublet Antennas. A typical all-wave doublet antenna system employing three doublet antennas is shown in Fig. 24. Doublet A_1 serves for the shortest wavelength, A_3 for the longest wavelength and A_2 for the intermediate wavelength. (Coils L_{A} serve merely to increase the effective length of doublet A_3 , improving reception at about 50 meters.) Although these three antennas work best at their own resonant wavelengths, they are sufficiently broad to give adequate pick-up at inbetween wavelengths. The connections between one or more of the doublets and the transmission line are transposed, as is done for A_1 and A_2 in Fig. 24, so that the signal voltages in the various doublets will add and give maximum transmission line current.

Whenever the wavelength of a desired radio signal is such as to give resonance for one of the doublet antennas in this system, the antenna impedance will be that of a normal doublet, which is 72 ohms. Since the transmission line is designed to have a surge impedance of about 72 ohms, a good match between the antenna and the transmission line is secured at resonant frequencies. When the desired signal does not give resonance for any one of the doublets, the impedance of the antenna will be much higher than 72 ohms, but there is one interesting characteristic of a transmission line which can be used to counteract this mismatch: A practical transmission line (only a few wavelengths long) has a high input impedance whenever its length is some multiple of $\lambda/2$ meters



FIG. 24. All-wave noise-reducing antenna made up of three doublets. Because of its general resembleoce to the web of a spider, this RCA-Victor product is commonly known as a spider-web antenna.

for a particular signal. By making the length of the transmission line a multi. ple of $\lambda/2$ for wavelengths in between the natural resonant frequencies of the three doublets, efficient transfer of signals from the antenna to the transmission line is obtained at in-between wavelengths as well. Whenever the manufacturer of an all-wave antenna specifies that the transmission line supplied with the system must not be cut, and that if too short it should be lengthened in equal lengths of a definite number of feet, you can be sure that the purpose is improvement of match at off-resonant frequencies.

With proper design it is possible to build a multiple-doublet all-wave antenna which is reasonably effective at

all wavelengths between the lowest and highest natural wavelengths of the doublets employed. Response at wavelengths below this range can be obtained by adding shorter doublets. Fairly satisfactory response will be obtained at wavelengths above this range since all of the doublets will then be shorter than $\lambda/2$, and will have a maximum current at their centers. At broadcast band wavelengths, however, this maximum current becomes quite low even for the longest doublet; to offset this, the special receiver coupling transformer shown in Fig. 24 is employed to convert the entire antenna system into a T antenna for broadcast band reception.

Observe that the midpoint of primary winding P is connected to ground through choke coil CH and condenser C. At broadcast band frequencies, very little signal current is fed to the transmission line by the doublets: the entire system acts as a T antenna, and signal currents which are picked up by the transmission line flow down both transmission line wires in the same direction to the center tap of P, then through CH (which has a low reactance at broadcast frequencies), through secondary winding S to the antenna terminal of the receiver, and through the receiver input coil to ground. Condenser C has such a high reactance at low frequencies that it acts essentially as an open circuit. When an all-wave doublet antenna system is converted to a T antenna in this manner, the system of course loses the noise-reducing and directional properties of a doublet.

Single All-Wave Doublets. An antenna arrangement which gives satisfactory reception on short-wave bands and which automatically converts to a T type antenna at broadcast frequencies without losing its noise-reducing properties and without sacrificing an impedance match is shown in schematic form in Fig. 25A. This system employs a single doublet antenna; it may be center-fed, in which case both sides of the horizontal doublet will be equal in length, or it may be fed from an offcenter point, in which case the sides of the doublet will be unequal in length.

This unique system employs two transformers at the antenna and two at the receiver. At high frequencies, condensers C_1 and C_2 become so low in reactance that they short out coils L_9-L_{10} and $L_{11}-L_{12}$, making the two right-hand transformers ineffective at high frequencies; the transmission line circuit therefore takes the form shown



FIG. 25. Taco all-wave noise-reducing antenna system, made by Technical Appliance Corp. Actual circuit is shown at A; the effective circuit for shortwave bands is at B, while the effective circuit for the broadcast band is at C. Dotted arrows show directions of noise currents at one instant of time; solid arrows show directions of desired signal currents at one instant of time.

in Fig. 25B at high frequencies (at short wavelengths). As you can see, this is a conventional doublet arrangement; if its length is $\lambda/2$ or less, current will flow along the entire antenna in the same direction at any instant of time. This current will flow through L_1 and L_2 in the same direction, inducing voltages in L_3 and L_4 which add together and cause current to flow down transmission line T_1 and up transmission line T_2 at one instant. The result is current flow through L_5 and L_6 in the same direction at any instant, inducing in L_7 a voltage which is applied to the input of the receiver. The

transformers used in this system provide the desired broad impedance match for efficient transfer of power over a wide range of high frequencies. Noise currents flow through leads T_1 and T_2 in the same directions, then in opposite directions through L_5 and L_6 to their common terminal and to ground, so no noise signal voltages are induced in L_7 .

At broadcast band frequencies, condensers C_1 and C_2 have high reactances, and hence have no shorting effect upon secondary windings L_9 - L_{10} and L_{11} - L_{12} . Furthermore, at these low frequencies coil elements L_1, L_2, L_3, L_4 , L_{B} , L_{B} and L_{T} have negligible inductive reactance and may be neglected, giving in effect the circuit shown in Fig. 25C. Condenser C_8 in Fig. 25A has a high reactance at broadcast band frequencies, and serves to prevent L_7 from shorting L_{18} . Again we find that noise signals picked up by the vertical down leads T_1 and T_2 are cancelled out at the receiver end of the transmission line (in primary windings L_{11} and L_{12}). The entire down lead system acts as a vertical or T antenna, with the flat top section serving merely as a load which produces a larger r.f. current at the point along the antenna to which coil L_8 is connected. This current flows down through L_8 , then divides to flow through L_9 and L_{10} and through L_{11} and L_{12} to the mid-point and then to ground, as shown by the dotted line arrows. This is normal current distribution for a T type antenna, but of course it does not feed the receiver. The flow of antenna current through winding L_8 serves to induce voltages in L_9 and L_{10} which act in the same direction and serve to circulate a signal current through the entire transmission line circuit. This current flows through windings L_{11} and L_{12} in the same direction, inducing a strong radio signal voltage in L_{18} . This winding in turn feeds the voltage to the receiver.

Use of Counterpoise for Noise Reduction. Another arrangement of a single doublet antenna which gives satisfactory signal pick-up along with noise reduction on all bands is that which employs a counterpoise for picking up strong noise signals. These noise signals are fed through the receiver input circuit in such a way that they "buck out" or cancel the usual noise signals, giving essentially noise-free reception on bands in which the doublet is acting as a T.

The circuit diagram of a typical antenna-counterpoise installation on a



FIG. 26. All-wave noise-reducing antenna system employing a counterpoise (RCA-Victor Master Antenna as used with RCA-Victor model 99K receiver).

three-band receiver is given in Fig. 26A. Each band has a separate receiver matching transformer as a part of the receiver preselector (an antenna system like this can be used only with receivers designed especially for it). For short-wave reception, band-changing switch SW_1 is at position C, and we have a normal noise-reducing doublet antenna with L_1 - L_2 serving as the receiver matching transformer. Condenser C_5 provides a path from the mid-point of L_1 to ground for noise signals, and 1-megohm resistor R_{26} allows static charges picked up by the antenna to leak off to ground gradually without causing noise.

For broadcast band reception, switch SW_1 is at position A, and we have T antenna action. To simplify an analysis of the antenna system under these conditions, we can redraw the circuit as in Fig. 26B to include only those parts which are effective during broadcast band reception. The length of the counterpoise is 10 feet longer than half the length of the transmission line, and consequently this lower portion of the transmission line (adjacent to the counterpoise) will pick up essentially the same amount of station and noise signals as the counterpoise. The upper portion of the transmission line is ordinarily quite high above the ground, out of the region of strong noise signals, but it will pick up and feed down through the transmission line a certain amount of noise along with the station signals. Let us trace the path taken by the noise signal currents, neglecting station signals for the time being.

Noise signals picked up by the T antenna will come down to point 1 in Fig. 26B (corresponding to point 1 at the center tap of L_1 in Fig. 26A; some will go through C_5 to ground, and the remainder will go through coil L_3 to point 2. Again the noise currents divide, with some going through L_7 in a downward direction to point 3, and others going through C_1 directly to point 3. From here the noise currents either go to ground through condenser C_{54} or take a return path up the counterpoise to the T antenna again. Noise signal currents coming down the counterpoise divide at point 3, with some solve to ground through C_{54} , some going directly to point 2 through C_1 , and some going through L_7 in the opposite direction to that taken by the T antenna noise currents. From point 2 these counterpoise noise currents go through L_3 to point 1, from which some go to ground through C_5 and others take the return path up the T antenna

and over to the counterpoise. By adjusting the value of trimmer condenser C_5 , the division of the counterpoise and T antenna noise currents throughout this circuit can be made such that the noise currents sent through L_7 by the two pick-up systems will be exactlyequal and opposite, and will therefore "buck out" or cancel each other.

In securing cancellation of noise signals in this manner, we also secure cancellation of any station signals which are picked up by the lower portion of the T antenna. This makes only the upper portion of the transmission line



FIG. 27. Tace variable-directivity short-wave antenns, with a selector control which gives the same effect upon directional characteristics as would rotation of a single doublet antenna.

available for picking up signals of broadcast band stations; coil L_3 and condenser C_1 serve to compensate for this short effective pick-up length, to compensate for the detuning effect of C_5 and to equalize the amount of energy transferred to L_6 at different broadcast band wavelengths.

Variable-Directivity Antennas

A short-wave antenna system which is designed to have controllable directional characteristics is shown in Fig. 27A. Two doublet antennas, each 60 feet long, are located at right angles to each other. At 37 meters they become $\lambda/2$ doublets and have maximum pickup. At wavelengths longer than 37. meters they are less than $\lambda/2$ but still have maximum current at the center and give quite satisfactory signal pickup. For wavelengths less than 37 meters they may become λ , $3/2\lambda$, 2λ , etc., antennas, with clover-shaped radiation patterns which make the directional adjustment more critical. Each doublet has a two-wire transmission line going to its own coil at the receiver end; the two coils are mounted at right angles to each other as shown in Fig. 27B, with

a third coil mounted in their center in such a way that it can be rotated for maximum coupling with either of the stationary coils or for any intermediate amount of coupling with both coils. In this way it is possible to control the direction from which maximum signal pick-up is secured. The performance is essentially the same as that of the rotatable beam antennas used by amateur radio enthusiasts and by commercial short-wave stations for long-distance radio communication.

TEST QUESTIONS

Be sure to number your Answer Sheet 33FR-2.

Place your Student Number on every Answer Sheet.

Never hold up one set of lesson answers until you have another ready to send in. Send each lesson in by itself before you start on the next lesson. In this way we will be able to work together much more closely, you'll get more out of your Course, and you will receive the best possible lesson service.

- 1. In addition to feeding strong signals from desired stations into a receiver, what other characteristic must a good receiving antenna have?
- 2. What two components does a radio wave have? ELECTRIC FIELD COMPORENT & MAGNETIC FIELD COMP. 3. When a radio signal produces a half-sinc-wave distribution of current along a receiving antenna, how many wavelengths long is the antenna? LONG.
- 2 WAVE LENGTH 4. What natural condition in space makes long-distance radio communication HPPIR ATMOSPHERE possible? ALAYER IN THE
- 5. Describe briefly how sky waves cause fading at receiving locations near the limits of reliable ground wave reception from a given station.
- 6. At what point along a half-wave receiving antenna is r.f. signal current at maximum? Cente
- 7. What information is given by the horizontal radiation pattern of a receiving antenna? Ellertwentsi
- 8. Why is an antenna transformer sometimes used to connect an antenna to a the transmission line? To match the antenna impedance of the transmission line
 9. What change would you make in a noise-reducing antenna installation if
- it failed to give a satisfactory signal-to-noise ratio?
- 10. What is the chief purpose of the grounded Faraday shield which surrounds some types of built-in loop antennas? a position of the horyontal p of the antimuma. Increase high

World Radio History



FIG. 13. This diagram shows how a length of fishline is used to make the pointer slide over the tuning dial and make the gang tuning condenser rotate when the tuning knob is turned.

radio serviceman will usually be called even for a mechanical trouble such as a broken dial cord.

Restringing a Dial Cord. The diagram in Fig. 13 also gives you the necessary information for restringing the dial cord of our superheterodyne receiver. After removing the old dial cord, tie one end of the new cord to the spring, bring the cord entirely around the system exactly as shown in the diagram, then tie the other end of the cord to the spring and cut off surplus cord.

Now tune in a local station whose frequency you know, then push the pointer over the cord at the frequency number of this station. There are spring clips behind the pointer for gripping the cord. If restringing is properly done, the pointer will now move to the correct dial number for whatever station you tune in.

Bent Rotor Plates. Defects in a gang tuning condenser are generally easy to locate and repair. For instance, one of the rotor plates may become bent so that it touches an adjacent stator plate. This shorts one of the tuning circuits, and the receiver no longer plays. In other cases, the short is only momentary, and causes noise in the loudspeaker as the receiver is tuned.

Sometimes you can hear the condenser plates scraping against each other if you listen closely while turning the tuning control knob. The remedy simply involves straightening the bent plate with a putty knife or a thin-bladed screwdriver.

Trimmer Condensers

On top of each section of the gang tuning condenser is a small adjustable device known as a *trimmer condenser*. The purpose of these two trimmer condensers is to take care of any small differences which might exist between the two sections of the gang tuning condenser.

When the two tuning condenser sections are correctly matched by adjusting the "trimmers," both sections will automatically be tuned correctly for reception of a desired station when the tuning dial pointer is at the station frequency.

TOWERS IN THE SKY

Above the dust and clamor here below These sentinels of mind and spirit rise; Deftly their fingers touch the vibrant skies Where winged words are passing to and fro. They dwell on high, where rarer currents flow---

Perhaps behold, in pity and surprise, The low estate of recompense we prize, The narrow round of things we seek to know.

O stalwart towers, unshakeable, serene, Make us more worthy of your office here! Refine our message, whatsoe'er it be; Attune our thoughts to listeners unseen. On far horizons may our words ring clear, Proclaiming there the truth which makes men free!

-GEORGE P. CONGER



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Study Schedule No. 34

For each study step, read the assigned pages first at your usual speed. Reread slowly one or more times. Finish with one quick reading to fix the important facts firmly in your mind, then answer the Lesson Questions for that step. Study each other step in this same_way.

1. Brief A.M. Review; Characteristics of o Frequency-Modulated

Signal - - - - - - - - - - - - - - - - Pages 1-9

Spend at least twice as much time as usual on this first study step, because in it is the basis for the entire frequency modulation system of broadcasting. Once you get a clear picture in your mind of what an f.m. signal is like, you'll find that f.m. systems are no more difficult than the ordinary a.m. systems you have previously studied. Answer Lesson Questions 1 and 2.

2. An Elementry F.M. System; Additional Requirements in an F.M.

System – – – – – – – – – – – – – Pages 9–13

Here is a brief survey of an entire f.m. system from the microphone through the f.m. transmitter and through the f.m. receiver. Simplified circuits and explanations are used so that you can concentrate upon getting a broad general picture first. Answer Lesson Question 3.

□ 3. The F.M. Receiver; The Preselector; The Frequency Converter Pages 13-18

The block diagram of a combination f.m.a.m. receiver in this section will prove a valuable reference guide for studying the remainder of this lesson, because it shows at a glance the relationships between the various stages of the receiver. Study each circuit slowly and carefully, referring to the circuit diagram each time a part is mentioned.

🗌 4. The I.F. Amplifier - - - - - - - - - - - - Pages 18-27

Careful, slow study is required here also. You'll learn how the same i.f. tubes can be used both for f.m. and a.m. signals, and you'll learn how the limiter removes amplitude variations from f.m. signals. Answer Lesson Questions 4, 5, 6 and 7.

□ 5. The F.M. Detector - - - - - - - - - - - - Pages 27-32

Here's another section in which careful study will be well worth while if you intend to prepare yourself for any job whatsoever connected with f.m. systems. Answer Lesson Questions 8, 9 and 10.

🗋 6. Modern F.M. Transmitters; The Future of F.M. - - - Pages 32-37

This bird's-eye view of an f.m. transmitter completes your background knowledge of f.m. systems. With this knowledge, you should have no difficulty in mastering the details of any f.m. receiver maintenance job once you have completed the N.R.I. Course.

7. Mail your Answers for this Lesson to N.R.I. for Grading.

8. Start Studying the Next Lesson.

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World Radio History

Frequency-Modulated Signals

Brief Review of Amplitude-Modulated Signals

A THOROUGH understanding of the characteristics of a conventional amplitude-modulated (a.m.) signal will help you to picture in your mind the characteristics of a frequency-modulated (f.m.) signal.

The signal peaks of an unmodulated r.f. carrier will all have the same amplitude, as shown at the left in Fig. 1A. When the r.f. carrier is amplitudemodulated with a single sine-wave audio frequency (such as that corresponding to a pure audio tone), the peaks of the r.f. signal will rise and fall in amplitude in accordance with variations in the audio signal, as shown at the right in Fig. 1A.

A dash-dash line drawn through the positive peaks of the modulated r.f. signal in Fig. 1A will give a curve known as the envelope, which has the same wave form as the original audio signal. (Another envelope having this same audio wave form is obtained by drawing a dash-dash line through the negative peaks, but at this time we need consider only the upper envelope. Both envelopes are shown for each signal in Fig. 1.)

Horizontal line XX at the level of the unmodulated carrier peaks will be the reference line for this envelope. Thus, we can say that the positive peak of the envelope goes above reference line XX, and the negative peak of the envelope goes below reference line XX. Furthermore, with 50% modulation as in Fig. 1A, the envelope peaks will have exactly one-half the amplitude of the unmodulated carrier peaks.

When the envelope peaks are equal in amplitude to the unmodulated carrier peaks, as in Fig. 1B, we have 100% modulation of the carrier.

When the envelope peaks are greater in amplitude than the unmodulated carrier peaks, the modulation percentage is higher than 100% and ν the envelope becomes distorted, as shown in Fig. 1C. When an over-



Courtesy General Electric Co.

A million-volt discharge of man-made lightning created a crashing roar which completely drowned out the musical program when this combination f.m.-a.m. receiver was tuned to a broadcast band a.m. station. When the set was tuned to an f.m. station carrying the same program, however, the music emerged from the loudspeaker clear and strong, with only a hardly noticeable static buzz in the background even though a million volts of electricity was dancing and sputtering a few fest away. Truly this is dramatic proof of the noisereducing characteristics of an f.m. system.

modulated signal like this is demodulated in a receiver, the resulting audio signal will have the same distorted wave form as the dash-dash envelope in Fig. 1C.

When the frequency of the audio signal is varied, the frequency of the envelope changes in a corresponding

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FIG. 1. Examples of amplitude-modulated signals.

Thus, if we double the fremanner. quency of the sine-wave audio tone used for our first three examples in Fig. 1, we secure for 100% modulation the condition shown in Fig. 1D.

When voice or music is transmitted, the modulation envelope becomes far more complex, resembling that shown in Fig. 1E. When this envelope is \sqrt{also} be accomplished by varying the

analyzed, it is found to have many different audio frequency components.

Side Frequencies. As you already know, an r.f. carrier signal which is modulated with a single fixed-frequency sine wave tone is equivalent to three different pure r.f. signals. One will have the assigned carrier frequency and constant amplitude. The other two, called side frequencies, will be respectively above and below the carrier frequency by the audio frequency value, and each one will vary in amplitude between zero and onehalf the carrier amplitude as the percentage modulation varies between 0 and 100%.

Here is an example. If the highest frequency we wish to transmit is 5000 cycles (5 kc.), and the r.f. carrier frequency is 1000 kc., the three pure r.f. signals will be 1000 kc. (the carrier), 995 kc. and 1005 kc. (the side frequencies). If the lowest frequency to be transmitted is 100 cycles (.1 kc.), the two side frequencies going out with the 1000-kc. carrier will be 999.9 kc. and 1000.1 kc.

If a complex audio signal having many frequencies in the range from 100 cycles to 5000 cycles is being transmitted, there will be a 1000-kc. r.f. carrier signal traveling through space along with side frequencies ranging from 995 kc. to 999.9 kc. and from 1000.1 kc. to 1005 kc.

The percentage modulation for a complex signal varies from instant to instant; it is 100% when the sum of all the side frequency amplitudes at a particular instant of time is exactly equal to the carrier amplitude. Overmodulation occurs whenever the sum of all the side frequency amplitudes is greater than the carrier amplitude.

Characteristics of a Frequency-Modulated Signal

Modulation of an r.f. signal can

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frequency of the r.f. signal in accordance with sound pressure, while keeping the r.f. amplitude essentially constant. This, basically, is frequency modulation.

Sound is an alternate compression and rarefaction of air particles. This means that air in the path of a sound wave is alternately being increased and decreased in pressure with respect to normal atmospheric pressure. When a sound wave in air is converted into its electrical equivalent by a microphone, we have a corresponding audio signal with positive and negative alternations, one alternation representing compression and the other rarefaction of air. The amplitude of an alternation depends upon the loudness of the original sound in air, and the frequency of the audio signal depends upon the pitch (frequency) of the original sound.

The Resting Frequency. With frequency modulation, the r.f. signal which is radiated by the transmitter is at an assigned r.f. value called the resting frequency or center frequency whenever no sound is being transmitted. The resting frequency thus corresponds to the instant when the air at the microphone diaphragm is at normal atmospheric pressure. This is illustrated graphically by time interval 1 in Fig. 2. (The resting frequency of an actual f.m. transmitter would be some value above 88,000,000 cycles per second, but for illustrative purposes we arbitrarily choose 5 cycles to represent the resting frequency in this diagram.)

Although the compression half of a sound wave cycle in air could cause either an increase or decrease in the frequency of an actual f.m. transmitter (depending upon the number of a.f. stages and phase-reversing transformers which are between the microphone and the output of the modulator), it is a convenient and common practice among radio men to associate compression with an *increase* in f.m. transmitter frequency. We will follow this practice.

Frequency Deviation. When an a.f.^V signal is fed into an f.m. transmitter, the *frequency* of the r.f. signal thus swings above the resting value in proportion to increases in air pressure from the normal atmospheric value (compression). Likewise, the transmitter frequency swings below the resting value in proportion to decreases in air pressure from the nor-

mal atmospheric value (rarefaction). The amount of this swing above or below the resting frequency (the amount by which the instantaneous r.f. value differs from the resting value) is known as the *frequency deviation*. We illustrate these-r.f. signal frequency changes in Fig. 2 for a sound having a square-wave form.

Square-Wave Example. Time intervals 2, 3, 4 and 5 in Fig. 2 represent two cycles of a weak sound having a square-wave form. During the first half cycle (interval 2), the air directly in front of the microphone is compressed, and the r.f. carrier frequency for this interval is therefore higher than the resting frequency. We have indicated this by showing 7 complete cycles for time interval 2 (two more than for the resting frequency in interval 1).

During the second half of the first sound cycle, we have rarefaction at the microphone, and the r.f. carrier frequency for this interval is *lower* than the resting frequency. We indicate this by showing 3 cycles for time interval 3 (two less than for interval 1).

In the second cycle of the weak sound, the process repeats itself, with the r.f. carrier frequency going above the resting value for time interval 4, and going below the resting value for interval δ .

In an f.m. system, doubling the sound pressure of the original sound doubles the deviation in transmitter frequency from the resting value. Time intervals 6, 7, 8 and 9 in Fig. 2 illustrate this for the square-wave sound under consideration.

During the first half cycle of this louder sound wave (interval 6), we have compression, and the r.f. signal frequency is twice as much above the resting frequency as it was for corresponding interval 2 of the weak sound. We indicate this in Fig. 2 by showing

9 complete cycles for time interval 6 (this is 4 cycles above the resting value).

For the rarefaction half of the loud sound cycle (interval 7), the r.f. signal frequency swings just as much below the resting frequency as it swung above the resting frequency for the compression half cycle. Thus, we show 1 cycle for time interval 7, this being 4 cycles below the 5-cycle value for the resting frequency.

The 4-cycle swing above and below the resting frequency is repeated for intervals 8 and 9 in Fig. 2, to give the second cycle of the loud sound. \checkmark

Sine-Wave Example. The squarewave audio signal in Fig. 2 showed how sudden changes in the amplitude of an audio signal affect the output signal frequency of an f.m. transmitter. Now let us see how gradual changes in amplitude, such as those occurring in the sine-wave audio signal shown in Fig. 3, will affect an f.m. transmitter. This diagram can either represent variations in air pressure from a normal atmospheric value (as in Fig. 2) or positive and negative a.f. signal voltage swings.

First of all, we can say that the r.f. signal will be at its resting value whenever the audio signal passes through zero, such as at points a, e and g.

As the audio signal increases in amplitude from a to b to c, the frequency of the r.f. signal will rise above the resting value in a similar manner, to a maximum r.f. value corresponding to peak amplitude c. As the audio signal decreases gradually in amplitude to zero again at point e, the r.f. signal frequency will drop gradually back to the resting value in a similar manner. Note particularly that for the entire interval of time from a to e when the a.f. signal is positive, the transmitter frequency is above the resting value. When the a.f. signal goes through its negative half cycle from e to f to g, the r.f. signal frequency will decrease in a correspondingly gradual manner from its resting value to a minimum value corresponding to point f, then rise gradually again to the resting value to complete the sine-wave cycle.

Complex Voice or Music Signals. When a complex audio signal like that shown in Fig. 3B is transmitted by an f.m. system, the transmitter frequency will vary above and below the resting value in accordance with the amplitude and polarity of the audio signal at each instant, even though this voice or music signal contains many different audio frequencies. f.m. as they are used today.

Theoretically, the full audio frequency spectrum with a full range of volume can be handled satisfactorily by an f.m. system regardless of how small or how large the maximum deviation value may be. In actual practice, however, the added requirements of a high signal-to-noise ratio and minimum inter-station interference at receivers make necessary a high value for the maximum frequency deviation. The greater the maximum frequency deviation employed for desired signals, the less noticeable to the listener will be a given frequency deviation due to an interfering signal.

The channels which have been made available to f.m. broadcast stations in



Thus, we can say that the r.f. signal frequency will be above the resting value whenever the audio signal is *positive*, with the deviation from the resting value being proportional to the positive amplitude at each instant. Likewise, the r.f. signal frequency will be *below* the resting value whenever the audio signal is negative, with the deviation being proportional to the negative amplitude at each instant.

Amount of Frequency Deviation. Keeping in mind the fundamental f.m. fact that the instantaneous deviation in the frequency of the r.f. signal is proportional to the instantaneous amplitude of the audio signal, let us now consider actual frequency values for the United States and its possessions by the Federal Communications Commission are .2 mc. apart in the very high frequency band between 88 and 108 mc., with the lower portion of the band from 88 to 92 mc. being reserved for non-commercial educational f.m. stations. Each channel assignment represents the assigned resting frequency of the station (the unmodulated signal frequency of the station).

A guard band at least 25 kc. wide is required by the F.C.C. beyond each extreme frequency swing of a station, which leaves a maximum of 75 kilocycles for the permissible frequency swing in either direction from the assigned resting frequency. Percentage modulation, as we know it in connection with the a.m. system of broadcasting, does not exist in an f.m. system. Since the greatest permissible frequency deviation for a loud sound is 75 kc., this deviation corresponds to 100%_modulation_of_ an f.m. transmitter. Over-modulation (a deviation greater than 75 kc.). will not cause distortion in receivers, however, unless it makes the signal sweep beyond the linear portion of the S curve for discriminator action (to be considered later in this lesson).

The monitor engineer in an f.m. studio has meters before him which tell when the loudness level at the microphone is exceeding the value which gives the full 75-kc. deviation. Whenever necessary, he reduces the gain of the studio a.f. amplifier enough to prevent the transmitter from swinging more than 75 kc. off from the resting frequency. Likewise, the monitor increases the a.f. gain when the loudness level at the microphone drops way down for appreciable periods of time. Thus, the goal of the monitor engineer is essentially the same, for both f.m. and a.m. systems: To keep the audio level at receivers as high as is practical without making the program sound unnatural and without causing overmodulation at the transmitter.

Actual Example. Suppose that an f.m. station is assigned the 97.3-mc. channel, and its microphone is picking up a loud 1000-cycle sound. The resting frequency of this station will be 97.3 megacycles, and the maximum permissible deviation will be 75 kc. (.075 mc.) above and below this resting value. Thus, the r.f. signal frequency will go up to 97.375 mc. for the positive peak of each audio cycle, and will drop down to 97.225 mc. for the negative peak of each audio cycle. Since the audio signal passes through 1000 complete cycles in each second,

the f.m. transmitter will go through 1000 complete swings (from 97.3 to 97.375 to 97.3 to 97.225 to 97.3) in . each second in order to follow the positive and negative amplitude variations of the audio signal.

If the lowest loudness level to be transmitted in this example is 1/100 of the loudest level, the deviation for this weakest audio signal will be .075 mc. divided by 100, or .00075 mc. For this weak signal, then, the r.f. signal frequency will vary from 97.30075 mc. to 97.29925 mc. and there will be 1000 complete swings like this in each second.

If the f.m. station in our example is stransmitting a loud 100-cycle audio signal, the deviation will still be 75 kc. on each side of the resting frequency, but now there will be only 100 complete swings back and forth between the two extreme frequencies in each second.

F.M. Side Frequencies. Mathematical computations as well as actual measurements tell us that the continually varying r.f. signal frequency for an f.m. transmitter is equivalent to a carrier frequency and an infinitely large number of side frequencies. Fortunately, however, the essential side frequencies needed for highfidelity reproduction are within the 150-kc. deviation range over which the r.f. signal sweeps. Receivers are designed for reception of this range, and consequently they receive the essential side frequencies.

Side frequencies more than 75 kc. off from the resting frequency might create interference with adjacentchannel stations if these stations were in the same locality, but the Federal Communications Commission prevents interference of this nature by keeping channel assignments at least 400 kc. (.4 mc.) apart for f.m. stations in the same service area. Technical Data on Side Frequencies. When the carrier of an f.m. transmitter is modulated with an audio signal having a constant frequency of f cycles per second, the r.f. signal swinging above and below the resting frequency is equal to the following fixed-frequency signals: A signal having the resting frequency of the transmitter; two side frequencies, one above and one below the resting frequency by the audio value f; two side frequencies, above and below the resting frequency by twice the audio value f; two side frequencies, above and below the resting frequency by three times the audio value f; etc.

Here is an example. If the resting frequency of an f.m. transmitter is 103.5 mc. and the audio modulation frequency is 10,000 cycles (.01 mc.), the equivalent individual frequencies will be: 103.5 mc., 103.49 mc. and 103.51 mc., 103.48 mc. and 103.52 mc., 103.47 mc. and 103.53 mc., etc., down to zero and up to infinity.

In your work with f.m. apparatus, however, you will find it more convenient to think of an f.m. signal as a single r.f. signal which is continually varying in frequency above and below its resting frequency. (We do this in amplitude modulation when we consider a modulated r.f. signal as a single r.f. signal which is varying in amplitude from instant to instant as in Fig. 1A.)

Multiplex F.M. Operation. Frequency modulation has the added advantage of permitting multiplexing, the sending of two or more programs or types of intelligence by the same transmitter, without increasing the required band width for the station and without interference between the programs being transmitted.

Here is an example: To transmit code messages along with a broadcast program, a 20-kc. oscillator signal (above the audio range) could be made to vary in amplitude in accordance with the code modulation, and this modulated 20-kc. secondary carrier then fed into the transmitter along with the regular voice or music program. At the receiver, the voice or music program would pass through the stages in the conventional manner, and the 20-kc. modulated code aignal would be taken out ahead of the audio amplifier, and demodulated by conventional means in a separate detector stage. Filters would be used to keep the 20-kc. signal out of the regular a.f. amplifier, to prevent overloading of a.f. stages.

How F.M. Systems Reduce Interference. At any instant of time, we can consider that an f.m. signal has a definite frequency, above or below the resting frequency of the transmitter. We can therefore represent a desired f.m. signal at the input of an f.m. receiver by a vector, as shown in Fig. 4A. The amplitude (length) of this desired-signal vector D will be constant, and the rate at which the



FIG. 4. Vector diagrams showing how an undesired signal U can affect the frequency and amplitude of a desired f.m. signal D.

vector rotates counter-clockwise will correspond to the frequency of the r.f. signal at the instant of time under consideration. The position of the vector with respect to the horizontal reference line in the diagram is unimportant.

Now, let us suppose that we also have at the input of our f.m. receiver an undesired signal. It may be due either to noise interference or to an undesired r.f. carrier signal from some other station, but nevertheless we know that this undesired signal has both frequency and amplitude. Let us say that at the instant of time under consideration for Fig. 4A, the



Construction. Each trimmer condenser consists of one fixed and one movable metal plate, separated by a sheet of *mica*, an insulating material. Tightening the adjusting screw brings the

movable plate closer to the fixed plate, and thus increases the electrical size or capacity of the trimmer. Loosening the screw reduces the capacity of the trimmer.

The construction of a trimmer condenser is so simple that this unit rarely becomes defective. Correct adjustment is the chief problem of the radio man when working with the trimmer condensers in a receiver.

Adjustment. Although the actual adjustment of a trimmer condenser with a screwdriver or wrench is an extremely simple job, considerable knowledge of radio servicing techniques and radio fundamentals is necessary in order to know when and how much to turn the screwdriver or wrench during the alignment procedure. This is another highly practical technique which you are going to learn in the N.R.I. Course.

I. F. Transformers

Also on top of the chassis base of our superheterodyne receiver are two parts which look like square metal cans. Through the two holes in the top of each can, we can see adjusting screws if we look carefully. These screws are fairly reliable identifying clues which tell that the parts are i.f.transformers. The square cans are simply shields which protect the inner coils from the effects of stray electric These shields and magnetic fields. also keep the fields of the coils from affecting other parts.

Construction. Figure 14 shows what you would see if you removed a few nuts, then lifted up the square shield of one of the i.f. transformers. Inside this shield are two small coils mounted on a wood rod. Above the coils are two trimmer condensers, one



Courtesy Western Electric Co.

The radio man at the right is adjusting a trimmer condenser in the transmitter unit of a two-way mobile police radio set, while the other man listens to the effect of the adjustment with a headset plugged temporarily into a test jack on the transmitter.

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undesired signal has the amplitude and phase represented by vector U in Fig. 4B.

When we combine undesired-signal vector U with desired-signal vector D, as shown in Fig. 4C, we obtain resultant vector R. The amplitude of vector R is greater-than that of desired-signal D, showing that an interfering signal can affect the amplitude of a desired incoming signal. Furthermore, vector R is ahead of vector D, indicating that the sudden arrival of this interfering signal makes desiredsignal vector D increase its rotational speed suddenly; this means that during the time interval in which vector D is moving to its new position corresponding to vector R, the frequency of the desired signal is higher than it, should be.

Interference vector U in Fig. 4B may have any phase relationship throughout the entire whatsoever range of 360°. It may be in phase with desired-signal vector D, as indicated by interference vector U_2 in Fig. 4D; it may be 180° out of phase, as indicated by interference vector U_4 in Fig. 4D; it may lead the desired vector by 90°, as indicated by U_8 , or it may lag the desired vector by 90°, as indicated by U_1 . These are the four extremes which could exist, and an analysis of them will take care of intermediate undesired-signal vector positions as well.

Careful study of the vector diagram in Fig. 4D reveals that when the interfering signal is *in phase* with the desired signal, it *increases* the amplitude of the resultant signal without changing its frequency. When the interfering signal is 180° out of phase with the desired signal, it decreases the amplitude of the resultant signal without changing its frequency. When the interfering signal leads the desired signal, the resultant will be speeded up in frequency, and the amplitude will generally also be altered. Finally, when the interfering signal lags the desired signal, the resultant will be slowed down in frequency and will likewise be altered in amplitude. We thus come to the conclusion that an interfering signal in an f.m. system can change both the *frequency* and the *amplitude* of the desired f.m. signal

You will learn later in this lesson that the limiter stage of an f.m. receiver reduces all incoming signals to a constant low amplitude corresponding to radius L in Fig. 4D. Since changes in amplitude due to noise vectors are all outside of this acceptance area, the limiter stage of a properly



FIG. 5. Simple f.m. transmitter circuit, employing a condenser microphone shunted across the tank circuit of a conventional tuned-grid r.f. oscillator. designed and adjusted f.m. receiver wipes out amplitude changes due to noise.

Any change in frequency due to noise will show up in the output of an f.m. receiver, for changes in frequency are not affected by the limiter section of the receiver, and are converted into corresponding changes in amplitude by the following frequency discriminator section. By using a high maximum deviation value, however, changes in the frequency of the desired signal due to an interfering signal in the receiver can be made negligibly small with respect to frequency changes due to the desired audio program.

Field tests have proved that the ratio of desired to undesired signals increases as the maximum frequency, deviation of the system is increased. Tests have also proved that with a maximum deviation of about 75 kc. (a total swing of 150 kc.), interfering signals become quite imperceptible at the loudspeaker when the desired signal is at least twice as strong as the undesired signal.

An Elementary F.M. System

The F.M. Transmitter. In beginning our study of actual f.m. systems, let us first consider one of the simplest possible transmitter circuits capable of producing frequency-modulated signals.

A conventional vacuum tube oscillator circuit employing a coil-condenser resonant circuit can be tuned to frequencies in either direction from a reference frequency without appreciably changing the amplitude of the output signal. By taking a simple oscillator circuit like this and shunting its resonant circuit with a condenser microphone in the manner shown in Fig. δ , a simple but effective f.m. transmitter is obtained. Sound waves apply to the movable diaphragm of the microphone a varying pressure which makes this diaphragm move alternately towards and away from the fixed plate. This changes the capacity of the condenser microphone in accordance with variations in sound pressure, thereby alternately raising and lowering the frequency of . the r.f. oscillator to give the desired frequency-modulated signal.

There are a number of drawbacks to this simplified f.m. transmitter circuit. The condenser microphone would have to be close to the oscillator, to prevent pick-up of stray signals by the microphone leads. Each condenser microphone would have to be designed to give a definite frequency deviation with its particular oscillator, and even with this precaution the amount of deviation would vary with the particular performer using the microphone. Undoubtedly these problems could be solved by engineers if necessary, but fortunately there are more practical means for securing frequency modulation of a carrier signal. One such scheme will now be considered.

Inductance-Tube Type of F.M. Transmitter. A vacuum tube circuit can be made to act like an inductance, simply by adjusting circuit voltages so that the r.f. current drawn by the tube will lag the r.f. voltage applied to the tube. (This scheme is widely employed in the automatic frequency



FIG. 6. This practical f.m. transmitter circuit employs an oscillator control circuit consisting of a vacuum tube which acts like an inductance shunted across the r.f. tank circuit. The effective inductance value varies from instant to instant in accordance with variations in the amplitude of the a.f. signal, thereby giving frequency modulation.

control circuits of some radio receivers.) By connecting across the coilcondenser resonant circuit of an r.f. oscillator the vacuum tube circuit which is acting like an inductance, and by making the inductance of this vacuum tube circuit vary at a desired audio rate, we secure frequency modulation.

An inductance-tube type of f.m. transmitter circuit is given in Fig. 6., Triode vacuum tube VT_2 is connected into a conventional tuned-grid r.f. oscillator circuit, with L and C forming its tank circuit. Pentode tube VT_1 serves as the oscillator control tube which acts like an inductance; its y plate is connected directly to terminal 2 of tank inductance L, and its cathode is connected to terminal 1 of this coil through r.f. by-pass condensers C_0 and C_4 and the grounded chassis.

Now let us see how oscillator control tube VT_1 can act as an inductance in shunt with tank circuit L-C. First of all, r.f. tank voltage e in Fig. 6 must be considered as the r.f. voltage source acting upon the oscillator control circuit. The two r.f. signal paths connected in parallel across r.f. voltage source e are the plate-cathode path of oscillator controle tube VT_1 and path R_1 - C_1 - C_2 - C_4 .

At radio frequencies, path R_1 - C_1 - C_2 - C_4 is essentially resistive (the reactances of all three condensers are low with respect to the resistance of R_1), and hence the r.f. current flowing over this path is *in phase with its r.f. source voltage e.* This r.f. current develops across condenser C_2 an r.f. voltage which *lags* the r.f. current, and hence lags r.f. voltage *e.* (The a.c. voltage across a condenser always lags the condenser current.)

The r.f. voltage across C_2 acts on the control grid of VT_1 , making the tube pass an r.f. plate current which is in phase with the applied r.f. grid voltage. The r.f. plate current drawn from L by the oscillator control tube therefore lags the r.f. voltage across L. This is exactly the same phase relationship as for an inductance load across L; the oscillator control tube thus acts like an additional inductance shunting the tank circuit, and serves to increase the frequency of the r.f. oscillator.

*When two inductances are in parallel, the combined inductance is less than that of the smaller inductance. Lowering the total tank circuit inductance raises the oscillator frequency. If one of the parallel inductances is reduced in value, the oscillator frequency will increase; if one of the parallel inductances is increased in value, the oscillator frequency will decrease.

The a.f. modulation voltage which is applied to the control grid of VT_1 through R_2 varies the transconductance of the tube, and hence makes the a.c. plate current vary at an audio rate. Consequently, the inductance of this tube also varies at an audio rate. eeThis in turn makes the frequency of the r.f. oscillator swing above and below its resting value at an a.f. rate, giving frequency modulation of the r.f. carrier without appreciable variation in the r.f. amplitude. When the a.f. signal is removed or drops to zero. the r.f. oscillator returns to its resting frequency, which is determined by the size of inductance L, the normal inductance of the oscillator control circuit, and the tank circuit capacity.

Resistor R_2 prevents the a.f. source from shorting the input of the inductance tube; it is really an isolating resistor.

By adjusting the initial C bias on the oscillator control tube (varying the ohmic value of R_c) and by monitoring properly the a.f. voltage fed into the oscillator control tube, the maximum deviation in frequency can be made any desired amount. The circuit of Fig. 6 is therefore a suitable signal source for an f.m. system which is to be employed in transmitting intelligence. The f.m. signal would be taken either from terminal 2 or from a link which is coupled to tank circuit inductance L.

The F.M. Receiver. Up to the point at which frequency-modulated signals are converted into amplitude-modulated signals, an f.m. receiver uses essentially the same circuits as a corresponding a.m. receiver. The a.f. amplifier and loudspeaker are likewise essentially the same for both systems. The important new action in an f.m. receiver is the conversion of the f.m. signal into the desired audio signal, so let us consider now the basic



principles involved in this conversion.

If a frequency-modulated r.f. signal is introduced into an L-C resonant circuit which is tuned slightly above the highest deviation frequency, the r.f. voltage developed across the resonant circuit will vary with the frequency of the induced f.m. signal. An ordinary resonant circuit like the L-C circuit in Fig. 7A is thus a simple means for converting an f.m. signal into an a.m. signal.

A portion of the resonant response curve for this L-C circuit is shown in Fig. 7B. By applying to this curve in the proper manner the f.m. signal shown at the bottom of Fig. 2 (having a square-wave audio modulation), we can get a graphical picture of how the *amplitude* of r.f. output voltage $V_{\rm O}$ varies with the *frequency* of the incoming f.m. signal.

Let us assume that when the f.m. transmitter is at its resting frequency (time interval 1 in Figs. 2 and 7C), the operating point is at 1 in Fig. 7B. The vertical distance from 1 down to the horizontal reference line then determines the amplitude of r.f. voltage $V_{\rm C}$ across the resonant circuit, so we show the r.f. output voltage for time interval 1 as an r.f. signal having this same amplitude and the same resting frequency value, as at 1 in Fig. 7D.

When the frequency of the f.m. signal increases to the value for time interval 2 in Fig. 7C, we move up to point 2 on the response curve, and thereby secure the output signal at 2in Fig. 7D. Likewise, the input signal frequencies for time intervals 3, 4, 5, 6, 7, 8 and 9 in Fig. 7C give the output signal amplitudes shown at 3, 4, 5, 6, 7, 8 and 9 respectively in Fig. 7D.

Note that the signal frequency is unchanged by the resonant circuit, but the *amplitude* of the output signal varies in proportion to the frequency deviations of the input f.m. signal. A dash-dash line drawn through the positive peaks of the r.f. output voltage in Fig. 7D gives the original square-wave audio modulation of Fig. 2, showing that we now have an amplitudemodulated r.f. signal which can be demodulated with a conventional amplitude detector circuit. The fact that our output signal also varies in frequency does not matter, for the amplitude detector circuit removes all r.f. components. V

A complete circuit capable of converting an f.m. signal into an audio signal is shown in Fig. 8; this circuit can appropriately be called an f.m. detector. The final i.f. amplifier stage is also shown, and employs a 6SK7 pentode, with a double-tuned i.f. transformer $(C_3-L_3-L_2-C_2)$ in its input circuit. Resistor R_L across the primary of this i.f. transformer provides the loading required to flatten the response over the 150-kc. range through which



FIG. 8. This diagram shows how a resonant circuit can be used to convert the f.m. output signal of the i.f. amplifier into an a.m. signal which can be demodulated by a conventional diode detector.

the f.m. signal is varying during loud audio loudness levels.

Plate current of the final i.f. tube flows through coil L_1 , inducing an f.m. signal voltage in coil L of the L-Cresonant circuit in which frequencyto-amplitude conversion takes place. The r.f. voltage across this L-C circuit therefore varies in amplitude in accordance with variations in the original audio signal amplitude. This amplitude-modulated r.f. voltage is demodulated by one section of the 6H6 tube, and the resulting a.f. output voltage appears across diode load resistor $R_{\rm D}$. Condenser $C_{\rm D}$ removes r.f. variations across $R_{\rm D}$, so that only the desired a.f. voltage is fed into the a.f. amplifier.

Additional Requirements in an F.M. System

The f.m. transmitter and receiver arrangements just described represent the absolute minimum required in an f.m. system. Transmitters must have some type of crystal control, so as to maintain the assigned resting frequency value within the limits prescribed by law, and receivers must also have additional features which insure high fidelity and operating stability, reduce noise and simplify tuning.

The remainder of this lesson will be devoted chiefly to practical f.m. receiver circuits, but basic principles of highly stable modern f.m. transmitters will be covered briefly at the end of this lesson text to make your f.m. training more complete.

Resonant circuits have proved unsatisfactory for f.m.-a.m. separating purposes, chiefly because the slopes of their response curves are not sufficiently linear for high-fidelity purposes. Instead, we find in modern f.m. receivers a variation of the discriminator circuit used in a.f.c. circuits. This discriminator circuit is simple to adjust and gives linear operation. It uses a pushpull circuit to convert frequency variation to amplitude variations at the i.f. frequency and then detects this signal.

In a practical f.m. receiver, an additional section called the limiter is required just ahead of the discriminator to restore constant amplitude, so that all amplitude rises due to noise or station interference will be removed.

The limiter is a vacuum tube amplifier which is so designed that all desired signals will cause plate current saturation. This action of the limiter stage also levels out the response of the r.f. system.

Since the limiter in an f.m. receiver delivers the same amplitude for both

weak and strong incoming signals, the loudspeaker volume level does not change when tuning from one f.m. station to another. This means that it is unnecessary to use an a.v.c. system in an f.m. receiver to prevent blasting or counteract fading. Most f.m. receivers do have a.v.c., but this is primarily to prevent extremely strong input signals from driving the grids of r.f. or i.f. amplifier tubes positive and thereby causing blocking or interference. If grid current were allowed to flow during positive grid voltage values, the

The F.M. Receiver

Types of F.M. Receivers. Radio receivers which are capable of receiving f.m. signals can be divided into three groups, as follows:

1. Sets designed exclusively for f.m. reception. These can be table models or consoles, for listeners who already have good a.m. receivers.

2. Combination f.m.-a.m. receivers, providing all-wave a.m. reception along with f.m. reception. These will usually be large console sets and may also have automatic record changers, television, facsimile or sound-recording features.

3. F.M. converters, which are simply f.m. receivers without audio amplifiers or loudspeakers. They provide f.m. reception when connected resulting distortion of the incoming f.m. signal would produce harmonics which could beat with oscillator harmonics and other station signals to produce interfering signals at the i.f. value for f.m.

Another requirement in a modern f.m. receiver is a tuning indicator (either a meter or a cathode ray tuning indicator tube). F.M. resonant circuits are so broad due to loading that the average person would have difficulty in tuning in an f.m. station by ear alone.

to the a.f. input terminals of an ordi-

nary a.m. receiver. All f.m. receivers should have high sensitivity in order to operate the limiter at saturation for the widest frequency swings of the weakest signal to be received, and hence superheterodyne circuits are invariably employed.

Typical circuits used in combination f.m. and a.m. receivers will now be analyzed.

Combination F.M.-A.M. Receiver

A block diagram of a typical combination a.m. and f.m. superheterodyne receiver is given in Fig. 9. The circuits now to be considered will be



FIG. 9. Block diagram of a typical combination f.m. and a.m. receiver.

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those employed in this particular receiver arrangement. V

Note first of all that there are two antennas, one for regular all-wave a.m. signal pick-up, and the other for ultra-high-frequency f.m. signal pickup. The band-selecting switch automatically connects the correct antenna to the receiver. (Fairly satisfactory results can be obtained with a single combination antenna on some sets, as you will later learn.)

For amplitude-modulated signals, the local oscillator will operate about 455 kc. above the incoming a.m. signal frequency, while for frequencymodulated signals, it will operate at 4.3 mc. below the incoming f.m. signal frequency.* The i.f. signal produced by the mixer-first detector (about 455 kc. for a.m., or 4.3 mc. for f.m.) passes through the first i.f. amplifier section to a switching circuit.

For f.m. reception, the signals are fed into a second i.f. amplifier section, which raises the signal level the required amount for the limiter stage. The constant-amplitude output of the limiter is fed into the discriminator, and the resulting a.f. output signal is fed directly into the input of the a.f. amplifier.

For a.m. reception, the output of the first i.f. amplifier section is fed directly into a conventional a.m. detector, and its a.f. output is likewise fed into the input of the a.f. amplifier. The tuning indicator, if used, would * Although 10.7 mc. is the recommended and widely used i.f. value for f.m. receivers at the present time, you may also encounter other values. Some f.m. receivers are capable of operating not only on the 88-108-mc. f.m. band, but on the older 42-50-mc. band as well. This low-frequency band is no longer in use. These sets particularly may have a different value of i.f., generally lower in frequency. The higher value now in use minimizes chances for interference between the various signals which are picked up by the receiving antenna or produced in the receiver by superheterodyne action.

be connected through the band-changing switch to the a.v.c. source for either the f.m. or a.m. channel. A common power pack serves all sections.

F.M. Receiving Antennas

The antenna for f.m. reception will ordinarily be a doublet of the correct length to give half-wave operation at an average frequency in the f.m. band. Since commercial f.m. stations are assigned to frequencies between 88 and 108 mc., a doublet which resonates at about 98 mc. will ordinarily be used. With proper design, its response will normally be broad enough to give satisfactory coverage of the entire f.m. band.

A frequency of 98 megacleyes corresponds to a wavelength of 3.06 meters (wavelength in meters is equal to 300 divided by frequency in megacycles). A half-wave f.m. doublet antenna will therefore have a total length of about 1.5 meters; multiplying this value by 3.28 gives about 5 feet as the total length for an average f.m. doublet receiving antenna.

The simplest possible f.m. doublet is that shown in Fig. 10A, in which the transmission line is connected to the center of the 5-foot long doublet. Since a half-wave doublet has a resistance of about 72 ohms at its center, a transmission cable having a surge impedance of approximately this same value would ordinarily be Slight mismatches can be used. tolerated, however; in fact, a cable with a surge impedance of about 100 ohms will make the antenna response characteristic more uniform over the entire f.m. band. The transmission cable is connected to the receiver through a matching transformer.

At very-high frequencies, line-ofsight paths give the best transmission of radio waves, but, fortunately, reliable f.m. reception is possible con-



FIG. 10. Types of f.m. receiving antennas. The heavy arrow in each case indicates the direction (along the surface of the earth) from which signals are best received. The symbol λ represents "wavelength."

siderably beyond the line-of-sight range from a transmitter. (With a line-of-sight path, it should be possible to see the transmitting antenna with a telescope on a clear day from the position of the receiving antenna.)

A compact very-high frequency antenna built into the console cabinet of an f.m. receiver will often give adequate signal pick-up of stations up to about ten miles away. Even a quarter-wave vertical antenna can sometimes be used when signals are strong.

In locations considerably beyond the line-of-sight limit, the position of the receiving antenna becomes important. The half-wave horizontal f.m. doublet should then be as high as possible, and broadside (at right angles) to the path over which f.m. signals are arriving.

In remote locations where all of the desired f.m. signals are coming from essentially the same direction, such as from a single large city, a reflector like that shown in Fig. 10B can be used with a conventional doublet to increase the signal pick-up from the desired direction. The doublet is always placed between the reflector and the transmitter.

The antenna arrangements shown

in Figs. 10C and 10D give even better signal pick-up at remote locations. In Fig. 10C, one section of the doublet has the usual length of $\lambda/4$ (one-quarter wavelength), while the other section is several wavelengths long. The long section is aimed directly at or slightly above the transmitter.

In the V doublet arrangement of Fig. 10D, both horizontal sections are equal, several wavelengths long, and aimed in the general direction of the transmitter.

Before mounting an f.m. doublet permanently, it is always a good idea to move the antenna toward or away from the transmitter a distance of about 5 feet, and tilt it at various angles while an assistant is checking signal strength at the receiver by watching the tuning indicator or using a meter, to find the position which gives maximum signal strength. Cancellation due to reflected waves arriving over different paths will then be a minimum.

The Preselector

A preselector having at least one r.f. amplifier stage is just as desirable in an f.m. superheterodyne receiver as in an a.m. set, for r.f. amplification

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ahead of the mixer-first detector in an f.m. receiver provides much needed extra sensitivity and minimizes converter noise. Just as with a.m. receivers, the higher-priced f.m. receivers will usually have an r.f. stage in the preselector. Less expensive f.m. sets will have a simple antenna transformer with tuned secondary, feeding directly into the mixer-first detector.

The circuit of a typical r.f. amplifier stage for a combination f.m.-a.m. receiver is shown in Fig. 11A. To simplify the diagram, only one allwave a.m. range is shown; additional contacts on the three sections of the band-changing switch and additional input and output r.f. transformers would be provided for other a.m. ranges.

When the three switches $(S_1, S_2 \text{ and } S_2)$ S_3) are set to position 1, as shown in Fig. 11A, the entire preselector is connected for f.m. reception. The halfwave f.m. doublet feeds its f.m. signal into the grid circuit of the 6SK7 r.f. amplifier tube through f.m. antenna matching transformer L_1 - L_2 , with secondary winding L_2 being tuned to the incoming f.m. signal frequency by gang condenser section C_1 . The amplified r.f. plate current of the 6SK7 tube flows through winding L_3 of the f.m. input transformer for the mixer, inducing in secondary winding L₄ a corresponding r.f. voltage. Gang condenser section C_2 tunes L_4 to resonance at the desired station frequency, and the resulting f.m. signal voltage across this resonant circuit is applied to the grid and cathode of the mixer-first detector tube. Automatic C bias for the 6SK7 tube is provided by R_2 and C_3 .

Setting the three switches in the preselector circuit of Fig. 11A to position 2 automatically connects the circuit for a.m. reception on one band. Switch S_1 disconnects the secondary

of the f.m. matching transformer from the r.f. amplifier tube grid, and connects the grid instead to the secondary of the a.m. antenna matching trans-The all-wave a.m. doublet former. feeds its a.m. signal to the r.f. amplifier tube through a.m. matching trans-_ former L_5 - L_6 , with secondary winding L_6 being tuned to resonance by gang condenser sections C_3 and C_1 in parallel. (The conventional condenser section for a.m. is split into two sections, C_3 and C_1 , so that one low-capacity section will be available for f.m. to give a favorable L-to-C ratio.) The r.f. output of the 6SK7 tube is fed to the mixer through a.m. input transformer L_7 - L_8 , with secondary winding L_8 being tuned by C_4 and C_2 in parallel.

When the switches are at position $\mathscr{2}$ for a.m. reception, a.v.c. voltage is applied to the r.f. and mixer tubes through L_6 and L_8 respectively. No a.v.c. is used on these two tubes during f.m. reception in this circuit, but engineers are now recommending that a.v.c. be provided for the r.f. amplifier tube of an f.m. receiver to prevent overloading of any tubes by extremely strong f.m. signals. The a.v.c. voltage for f.m. would be supplied by the *limiter* section.

A variation of the preselector circuit of Fig. 11A which makes possible the use of one doublet antenna for both f.m. and a.m. reception is shown in Fig. 11B. The remainder of the preselector circuit is not shown, since it is identical with that in Fig. 11A. The connections are such that doublet antenna action is secured for f.m. reception, but the antenna is automatically converted into a plain vertical antenna for a.m. reception. Resonant circuit L-C is tuned to the mid-frequency of the f.m. band (about 98 mc.); since it is a series resonant circuit, it has a low impedance at or near resonance, and has the


FIG. 11. Typical preselector circuit arrangements for a combination f.m.-a.m. receiver. If the r.f. amplifier tube is to have a.v.c., terminal a of f.m. coil L_2 is connected to terminal b of a.m. coil L_3 instead of to ground. A switch at the two a.v.c. sources (shown in Fig. 20) insures that the correct a.v.c., voltage (for f.m. or for a.m.) will be fed to this tube.

effect of grounding the center tap of winding L_1 . Under this condition, the antenna acts as a very-high-frequency doublet for the f.m. band.

For a.m. bands, usually below 20 megacycles, the series L-C circuit is considerably off resonance, and hence has a high capacitive reactance. Winding L_1 is therefore ungrounded, and signal currents picked up by the two sections of the doublet flow through the two sections of L_1 to the center tap, then through winding L_5 to ground (L_5 has much lower reactance than the L-C circuit). The a.m. signal current through L_5 induces a corresponding r.f. voltage in L_6 , and this is fed through the r.f. amplifier tube to the mixer in exactly the same manner as for the circuit of Fig. 11A.

Observe that a resistor (R)' is shunted across each f.m. resonant circuit in Figs. 11A and 11B. These resistors serve to load the resonant circuit, providing broad resonance which gives essentially uniform response at all frequencies within the deviation range of an incoming f.m. signal. Ordinarily, resistor R has an ohmic value somewhere between 10,-000 and 20,000 ohms, with the higher value being preferred in order that the preselector will have good selectivity. When no r.f. amplifier stage is used, a band-passed preselector circuit ahead of the frequency converter is highly desirable, to broaden the response and at the same time improve selectivity.

The Frequency Converter

In any superheterodyne receiver, the local oscillator and the mixerfirst detector constitute a section commonly called the *frequency converter*. The local oscillator feeds into the mixer-first detector an unmodulated r.f. signal which beats with the incoming modulated signal to produce the correct i.f. signal.

A number of different circuit arrangements can be employed in the frequency converter of a combination f.m.-a.m. receiver. Some designers will use a pentagrid converter tube, some will use a combination triodepentode tube, and some will use separate tubes for the oscillator and mixerfirst detector.

It is customary to operate the local oscillator *above* the incoming signal frequency for f.m. reception, to secure greater stability and prevent local oscillator signals from creating interference in the television band below 88 mc. On a.m. bands also the oscillator is usually operated *above* the incoming signal frequency. connected across each coil. These are the trimmer condensers which you adjust with a screwdriver through the holes in the top of the shield during alignment of the receiver.

What I.F. Transformers Do. When the gang tuning condenser and its coils are adjusted or tuned to a desired incoming carrier signal, these parts cannot always keep out undesired signals having different frequencies. I.F. transformers are much more selective, however.

The only signal which can get through a properly adjusted i.f. transformer without serious reduction in strength is the one whose carrier frequency is the i.f. value of the receiver (456 kc. in this case).

The oscillator-mixer-first detector automatically changes the carrier frequency of the desired incoming signal to the i.f. value, but changes undesired signals to other frequencies which cannot get through. One i.f. transformer transfers the desired signal from the mixer-first detector to the i.f. stage, and the other i.f. transformer transfers the desired signal from the i.f. stage to the second detector.

Open Coils. The coils in an i.f. transformer are made from small-size insulated copper wire. This wire sometimes breaks due to vibration or jarring, to strains set up by changes in temperature, to corrosion at soldered joints, or to swelling of the coil form in damp weather. Any break in the coil wire interrupts the signal circuit, thereby causing complete failure of the receiver. If the break is at a joint, it can sometimes be repaired by soldering. If not, a new i.f. transformer must be installed.

Sometimes a connection may be weakened by corrosion without being completely open. This defect can



FIG. 14. This is what you see inside the i.f. transformer when you lift the metal shield can.

cause loud crackling noises in the loudspeaker as the connection opens and closes at irregular intervals.

These are just a few of the many receiver troubles which cannot be seen by examining the parts. A systematic and efficient trouble-locating procedure such as you learn in the N. R. I. Course is absolutely necessary for locating defects like open i.f. transformers.

Power Transformer

We now have only one device left to consider above the chassis base of our superheterodyne receiver. This is the *power transformer*, a large and heavy unit located at the right of the loudspeaker when looking at the back of the chassis. Since all of the leads of this radio part are located underneath the chassis, however, let us postpone its study until we turn the chassis base upside down. Switching sections must be provided on the band-changing switch of a combination f.m.-a.m. receiver to insert the correct inductance and variable capacity in the oscillator tuning circuit of the frequency converter. The oscillator would thus be tuned 10.7 mc. (or whatever other i.f. value is used for f.m.) above the incoming signal frequency for f.m. reception, and about 455 kc. above the incoming signal frequency for a.m. reception.

An example of a conventional frequency converter circuit employing a type 6K8 combination triode-pentode tube is given in Fig. 12. This circuit could be connected directly to the



FIG. 12. Typical frequency converter circuit for a combination f.m.-a.m. receiver.

preselector circuit of Fig. 11, with all tuning condenser sections being controlled by one tuning knob and all switches being controlled by the bandchanging knob.

When switches S_4 and S_5 in Fig. 12 are at position 1 for f.m. reception, oscillator tank circuit inductance L_1 is tuned 10.7 mc. above the incomingf.m. signal frequency by tuning condenser section C_5 . The high-frequency trimmer is C_8 , and the lowfrequency padder is C_4 , while C_1 serves as a temperature-compensating (Condenser C_1 changes condenser. its capacity in a definite manner with thereby minimizing temperature, oscillator frequency drift by compensating for other capacity changes which occur as the set comes up to normal operating temperature after being turned on.)

Feed-back from the oscillator plate circuit is provided by inductance L_2 . Automatic C bias for the oscillator is provided by-the flow of grid current through grid resistor R_2 and grid condenser C_{11} . D.C. blocking condenser C_{10} prevents the plate supply voltage of the oscillator triode from sending direct current through oscillator feedback coil L_2 to ground.

All leads in the f.m. oscillator section of the circuit should be as short as possible, so that resonance effects will be localized to the desired L_1 - C_5 tuning circuit. All grounded elements in a section should be connected to a common point on the chassis. \mathbf{At} leads very-high frequencies, long have appreciable inductance and capacity to the chassis, and hence might form additional resonant circuits which create trouble.

When switches S_4 and S_5 in Fig. 12 are at position 2 for reception on one of the a.m. bands, the oscillator tank circuit inductance is L_4 . The tank circuit is tuned 455 kc. above the incoming a.m. signal frequency by tuning condenser sections C_5 and C_6 in parallel. C_7 serves as the high-frequency trimmer, and C_8 serves as the low-frequency padder. Feed-back energy from the plate circuit is now supplied by L_3 , for removal of the short across L_3 by S_5 allows r.f. plate current to flow through both L_2 and L_3 to ground.

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The I.F. Amplifier

We cannot use a standard 455-kc. i.f. value for both f.m. and a.m. reception. The gain of a 455-kc. i.f. amplifier drops entirely too much when it is band-passed sufficiently to pass a 150-kc. wide f.m. signal. Furthermore, the selectivity of this bandpassed 455-kc. i.f. amplifier would be way too poor for standard a.m. reception. A 10.7-mc. i.f. amplifier is satisfactory for f.m., but its selectivity is likewise too poor for a.m. reception.

To meet these conflicting requirements in a combination f.m.-a.m. receiver, separate i.f. transformers can be used for f.m. and a.m., so that a.m. signals will pass through standard 455-kc. i.f. transformers, and f.m. signals will be served by 10.7-mc. i.f. transformers. With the exception of the first i.f. transformer primary, the connections can be such that signals will take the correct transformers in traveling through the i.f. amplifier. It is even possible to mount both the f.m. and a.m. transformers for a given stage in the same shield can.

When corresponding windings of the i.f. transformers for f.m. and a.m. are connected in series as shown in Fig. 13, signals will automatically take the correct path. Thus, when a 455-kc. a.m. signal is coming through, the transfer of signals will be accomplished by the 455-kc. a.m. transformer L_3 - C_3 - L_4 - C_4 , because L_1 will be essentially a short-circuit path. When 10.7-mc, f.m. signals are coming through, signal transfer will be through f.m. transformer L_1 - C_1 - L_2 - C_2 , because C_3 will be essentially a shortcircuit path. In each case, the i.f. transformer not in use will be so far off resonance that its effect will be negligible.

, $\overline{I.F.}$ Switching. During broadcast band a.m. reception, the local oscillator produces a signal somewhere between about 1000 and 2000 kc. (1 and 2 mc.) when the local i.f. value for a.m. is 455 kc. The sixth harmonic of the oscillator will then cover 6 to 12 mc., and there will be one tuning dial position at which this oscillator signal coincides with the 10.7-mc. i.f. value for f.m. With the series arrangement of 455kc. and 10.7-mc. i.f. windings, this oscillator harmonic (if strong enough) could get through the first i.f. transformer for f.m., and drive the self-bias so far beyond cūt-off (due to increased plate current through the cathode resistor) that the desired 455-kc. i.f. signal would be blocked. The result would be a "dead spot" in the broadcast band at the setting which made this oscillator harmonic equal to 10.7 uc.

Other harmonics of the local oscillator can likewise create dead spots at different broadcast band dial settings,



FIG. 13. I.F. transformer arrangement suitable for a combination f.m.-a.m. receiver. Separate i.f. transformers with corresponding windings connected in series are used for f.m. and a.m. No switches are employed, for desired signals automatically take the correct path due to circuit impedance values.

and hence it is necessary either to disconnect or short out the 10.7-mc. first i.f. transformer during broadcast band reception. When a switch is provided at the i.f. amplifier input for this purpose, only the desired i.f. signal passes through the first i.f. tube to the other i.f. transformers, and hence no additional i.f. switches are needed.

An example of a complete i.f. amplifier section which has provisions for disconnecting the unused first i.f.transformer primary is shown in Fig. 14. Section S_1 of the band-changing switch connects to the converter the correct i.f. transformer primary winding $(L_1 \text{ or } L_7)$ for the particular signals being received. All other pairs

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of i.f. windings are still connected in series just as in Fig. 13, however.

The i.f. input switch can just as well be between the grid of the first i.f. amplifier tube and the secondary windings of the first i.f. transformers, or can be used to short out the primary or secondary winding of the unused first i.f. transformer.

Note that the i.f. transformers for a.m. have fixed condensers, with tuning being accomplished with adjust-Pulverable iron-core inductances. ized iron cores are not effective at the 10.7-mc, i.f. value for f.m., but at a high frequency like this a low-resistance non-ferrous core (such as copper, aluminum or brass) can be used instead to vary the inductance. Nonferrous cores of this type are inserted from the grounded end of the i.f. coil; they vary the inductance due to eddy currents induced in the core, and also add capacity to the tuning circuit. Non-ferrous cores can also be used for r.f. coils in f.m. circuits.

Sometimes the final i.f. stage is used only during f.m. reception. A simple method for eliminating this stage during a.m. reception involves grounding point a in Fig. 14 permanently, and connecting secondary $L_{10}-C_{10}$ of the second 455-kc. i.f. transformer directly to the a.m. second detector. Transformer L_{11} - C_{11} - L_{12} - C_{12} would be omitted.

Both i.f. amplifier tubes in Fig. 14 have resistors in their cathode leads to provide automatic C bias for both f.m. and a.m. reception. During a.m. reception, however, the diode a.m. detector supplies an a.v.c. voltage to the first i.f. amplifier tube, as well as to preceding tubes. (During f.m. reception, the a.v.c. voltage is provided by the limiter stage.)

The Limiter. The purpose of the limiter section in an f.m. receiver is to remove amplitude variations from the f.m. signal at the output of the i.f. amplifier, so that the signal fed into the discriminator will have constant amplitude. In serving this purpose, the limiter automatically corrects deficiencies in the frequency response of the preceding r.f. and i.f. stages. The limiter also provides an a.v.c. voltage for use during f.m. reception.

A typical single-tube limiter stage is shown in Fig 15. To understand how the 6SJ7 pentode in this circuit can function as a limiter, we must first consider the E_g - I_p characteristic curves for this tube under various



FIG, 14. Circuit diagram of a complete i.f. amplifier for a combination f.m.-a.m. receiver. Note the use of a switch for the primary windings of the first i.f. transformers. The groups of three parallel lines with arrows on a.m. transformers represent adjustable pulverized iron cores; these are adjusted in the same way as ordinary i.f. trimmers during i.f. alignment.

operating conditions.

When a 6SJ7 pentode is operated at normal voltages for amplifying purposes, such as with a 250-volt d.c. plate voltage and a 100-volt d.c. screen grid voltage, the static E_g - I_p characteristic of the tube will be like curve 1 in Fig. 16. This curve is essentially linear up to the highest plate current which can safely be passed by the



FIG. 15. Simplified one-tube limiter circuit for an f.m. receiver. Condenser C4 is provided for alignment purposes, as will be explained later.

tube, and consequently the plate current increases in proportion to positive grid voltage values. This is of no value for limiter action; we desire a characteristic which will make the plate current essentially constant regardless of how much the grid is driven positive, and hence the tube must be operated in such a way that plate current saturation occurs at a low positive grid voltage value.

When the d.c. plate and screen grid voltages of the 6SJ7 pentode are about 60 volts, we secure the desired condition whereby saturation begins at a fairly low positive grid voltage value, as indicated by curve 2 in Fig. 16. This curve is for the tube alone, and would be obtained from the circuit of Fig. 15 only if resistor R and condenser C were shorted out so as to give static conditions.

If a strong sine wave signal is applied to the grid of the 6SJ7 tube when it is operated with characteristic curve 2 in Fig. 16, the operating point will be at o, negative peaks of the input signal will swing beyond cut-off (beyond c), and positive peaks will swing into the saturation region of the curve (beyond point x at which saturation begins). As a result, both the positive and negative peaks of the plate current will be cut off, and we will secure a certain amount of the desired limiting action on strong signals.

Now let us return to the basic limiter circuit in Fig. 15, and consider its operation first with resistor R in the grid circuit but with condenser C removed. First of all, we can see that during half cycles which swing the grid negative, there is no grid current flowing through resistor R, and consequently the circuit acts essentially the same as if resistor R were shorted. In other words, our characteristic curve with R in the circuit will be the same for negative grid voltages as it was without R. For half cycles which make the grid swing positive, however, the flow of grid current through R will develop across it a voltage drop which opposes the applied positive grid voltage, and which consequently serves to provide a negative bias which



FIG. 16. Static characteristic curves of a vacuum tube for high (1) and low (2) plate voltages, and dynamic characteristic curve (3) for a limiter circuit using a low plate voltage and a resistor R (but no condenser C) in the grid circuit. The plate load resistance for the static curves is assumed to be equal to the resistance of the plate load resonant circuit at resonance during dynamic operation.

is at each instant proportional to the positive applied voltage. The characteristic curve for this condition will

be like curve 3 in Fig. 16, which is flatter than curve 2. The saturation effect is considerably more pronounced now, and plate current during positive half cycles is limited by the negative bias across R as well as by the saturation characteristic of the tube at the low d.c. operating voltages used.

With R alone in the limiter circuit, the bias developed across it would follow individual r.f. positive peaks, and would consequently be varying continually. In a practical limiter circuit, resistor R in Fig. 15 would be shunted by condenser C, with the time constant of R and C being equal to the time of several r.f. cycles. Un-



FIG. 17. These diagrams will help you to understand the action of the basic one-tube limiter circuit in Fig. 16.

der this condition, the R-C circuit assumes a definite negative bias voltage which is maintained relatively constant over several cycles of the r.f. input voltage, and which hence exists for negative half cycles as well as for positive half cycles.

An automatic bias of this nature gives a characteristic curve which will be somewhere between curves 2 and 3 in Fig. 16; with R and C large in electrical size so as to give a long time constant, the dynamic characteristic curve will approach static curve 2 because we are approaching a fixedbias condition, but with low electrical values for R and C, the dynamic characteristic will approach curve 3 because the bias will now vary almost instantaneously with input signal strength.

A typical dynamic characteristic curve for a single-tube limiter stage is shown in Fig. 17. With this characteristic, the bias voltage developed across R and C will be proportional to the average signal strength over several r.f. cycles, and consequently the operating point might be at a, b, c, or even beyond cut-off at d, depending upon the r.f. signal strength. Thus, with a sine wave input voltage e_{g} which places the operating point at point b, the wave form of the resulting plate current i_p will be as shown at C in Fig. 17. This plate current is far from being sinusoidal like the input voltage, but we will still secure a sinusoidal output voltage across plate resonant circuit L_3 - C_3 in Fig. 16 because this circuit is tuned to the 10.7 mc. i.f. value for f.m. It has a high Q factor, and therefore responds only to the desired fundamental frequency of the plate current pulses, rejecting the harmonics.

With the dynamic characteristic curve of Fig. 17 for a limiter stage, two things happen when the strength of the input signal increases: 1. The increased negative bias makes negative half cycles swing beyond cut-off for a longer period of time during each cycle, thereby reducing the operating angle during which plate current does flow; 2. The amplitude of the plate current pulses increases slightly, because an automatically produced bias can never completely counteract an increase in signal strength. The increased positive voltage on the grid produces a slight increase in the amplitude of the plate current pulses because the characteristic curve rises slightly in the saturation region, rather than being perfectly flat.

It is possible to design a limiter cir-

cuit so that for any reasonable increase in the r.f. input voltage to the limiter, the operating angle for plate current will decrease just enough to counteract the increase in the amplitude of the plate current pulses. The energy fed into plate resonant circuit L_3-C_3 in Fig. 15 at the fundamental intermediate frequency for f.m. will then be constant, and the desired i.f. output voltage across this limiter resonant circuit will likewise be constant.

When this goal in limiter circuit design is realized, the over-all dynamic characteristic curve for the limiter will be like curve 1 in Fig. 18A, in which the r.f. output voltage of the limiter is plotted against the r.f. input determine which over-all characteristic will be obtained, and hence these two parts in a limiter circuit are highly important.

Since the d.c. voltage produced across limiter grid resistor R is proportional to the strength of the f.m. signal at the limiter input, this d.c. voltage can be used for a.v.c. purposes during f.m. reception.

A limiter must be fast-acting (must have a short time constant) if it is to block out sudden noise surges. This assumes that there is sufficient r.f. amplification in the f.m. receiver so that the weakest desired signal will swing the limiter grid beyond the point at which saturation begins.

To show how the limiter can flatten



FIG. 18. These curves show how the limiter can correct the over-all response characteristic of the entire r.f. system (including the i.f. amplifier and the receiving antenna system) in an f.m. receiver.

voltage to the limiter. With this characteristic, the i.f. output voltage remains essentially constant regardless of input voltage, for all input voltage values which reach saturation (swing beyond point s in Fig. 18A).

If the design of the limiter circuit is such that the operating angle decreases faster than the plate current pulse amplitude increases, we have over-compensation and secure the over-all characteristic represented by curve \mathcal{S} in Fig. 18A. Likewise, if the amplitude of the plate current pulses increases faster than the operating angle decreases, we have under-compensation and secure the over-all characteristic represented by curve \mathcal{S} in Fig. 18A. The values employed for R and C in the limiter circuit of Fig. 15 the response characteristic of the r.f. system, let us assume that the limiter stage under consideration has a flat over - all grid voltage - plate voltage characteristic curve like 1 in Fig. 18A. Assume further that under this condition, all limiter input signals having peaks higher than 5 volts cause plate current saturation. (Positive grid swings will then go beyond point s, at which saturation begins.)

Now suppose that the r.f. system ahead of the limiter has the sharply peaked response characteristic shown in Fig. 18B. The vertical scale gives peak amplitude values; thus, if a strong input signal gave a 20-volt output peak amplitude at the resting frequency, this peak amplitude at the limiter input would drop down to 5 volts during the maximum deviation of 75 kc. from b to a to c). With the limiter characteristic of Fig. 18A, all signal peaks above 5 volts would be cut down to 5 volts by the limiter, and the solid-line curve in Fig. 18C would then represent the response characteristic of the r.f. system and limiter combined, for a strong f.m. input signal. Since this is flat over the desired band width of 150 kc., it is obvious that the limiter has flattened the highly-peaked response of the r.f. system.

Now suppose the receiver is tuned to a weaker f.m. signal, which gives a



FIG. 19. Typical one-tube limiter circuits.

resting-frequency peak of only 10 volts at the output of the r.f. system. The dash-dash response curve in Fig. 18B would then give peak values for various deviations. When this weaker signal is fed into the limiter, the output response would be as shown by the dash-dash line in Fig. 18C. The band width over which we have uniform response is now obviously insufficient for standard f.m. signals, and distortion will occur during loud portions of the program.

On the other hand, if the r.f. system has the broadly-peaked response characteristic shown in Fig. 18D, a signal having a 20-volt peak would give a combined r.f.-limiter response corresponding to the solid line in Fig. 18E after passage through the limiter. This response is ideal, being flat for a frequency swing of about 250 kc. If the

input signal should now drop to a maximum peak value of 10 volts, the input and output response characteristics would be as shown by the dashdash lines in Fig. 18D and 18E; even at this lower signal level, however, the combined r.f.-limiter response is_still. flat over a wide enough range to allow the entire frequency swing of 150 kc. to exist at the limiter output without changes in amplitude.

We thus arrive at the important conclusion that a broad over-all response for the r.f. and i.f. sections enables the limiter to handle weaker signals satisfactorily. Almost any sort of peak response is permissible if the r.f. and i.f. sections have sufficient gain, however, for the limiter will then flatten the over-all response of the preceding sections over the entire range of deviation frequencies.

Time Constant of Limiter. The time constant of the R-C circuit in the limiter is ordinarily made equal to the time of a few r.f. cycles, so the limiter will respond to general changes in the amplitude of the incoming signal without actually following individual r.f. cvcles.

The importance of having a fastacting limiter can be made clear by considering an f.m. signal which is varying in frequency from 75 kc. below to 75 kc. above its resting frequency (maximum deviation, corresponding to maximum loudness). With a receiver having a sharply resonant *x.f.* response like that in Fig. 18B, and with a peak limiter input of 20 volts at the resting frequency, the amplitude of the signal fed into the limiter will vary from 5 volts (at points b and c) to 20 volts (at a).

With the 20-volt input, the limiter will probably be operating near plate current cut-off, but this cut-off bias would be far too great to give the desired constant output amplitude when the limiter input drops to 5 volts. In order to keep the limiter output amplitude constant over the entire deviation range of 150 kc., the C bias should automatically reduce itself as the limiter input signal amplitude drops.

Noise surges which enter an f.m. receiver are oftentimes strong but of extremely short duration. If these surges are to be blocked out, the limiter must be able to increase its own negative bias automatically for the duration of each surge. This action can occur only if a fast-acting limiter (with a short time constant for R and C) is used.

The C bias produced by the R-C grid network of the limiter stage can change fast enough to flatten the r.f. response and squelch noise only if its time constant is kept very short (within the time of a few r.f. cycles).

Fast-Acting Limiter Circuit. In Fig. 19A is a practical limiter circuit employing a grid resistance $(R_1 + R_2)$ between the grid and cathode of the limiter, with condenser C_1 and the grid-cathode capacity of the tube acting with the grid resistance to provide the required time constant for fast limiter action. This is attained by using a low capacity value for C_1 and a low ohmic value for the grid resistance. A portion of the d.c. voltage developed across the grid resistance (the d.c. voltage across R_2) is used for a.v.c. purposes during f.m. recep- R_3 and C_2 form the convention. tional a.v.c. filter which keeps r.f. components out of the a.v.c.-controlled tubes.

Dual-Action Limiter Circuit. In the limiter circuit arrangement shown in Fig. 19B, both C_1 and C_2 have low reactances at very-high frequencies. R_1 and R_2 in series form the grid return path. R_1 and C_1 together have a time constant equal to a few r.f. cycles, and hence provide a rapidly changing bias equivalent to the fast limiter action of the circuit of Fig. 19A. R_2 and C_2 have a much longer time constant, and act to change the C bias voltage in accordance with the average strength of an incoming f.m. signal. In other words, R_2 and C_2 take care of the major changes in signal amplitude such as those occurring when tuning from a weak distant f.m. station to a strong local f.m. station, while R_1 and C_1 take care of changes in signal amplitude due to a peaked r.f. re-



FIG. 20. One-tube limiter circuit with tuning indicator. I.F. amplifier connections to the conventional diode detector for a.m. reception are also shown.

sponse characteristic or to noise interference. This circuit arrangement thus provides independent control over two of the important factors which affect the design of a limiter circuit.

Limiter with Tuning Indicator. A more complete single-tube limiter circuit is shown in Fig. 20. This arrangement includes an electric eye for tuning purposes during both a.m. and f.m. reception. The output circuit of the i.f. amplifier is also shown, and is identical with that in Fig. 14. The limiter input circuit is the same as that shown in Fig. 19A.

The a.v.c. voltage developed across R_2 in Fig. 20 is fed to the tubes which are to be a.v.c.-controlled during f.m. reception through a.v.c. filter R_8 - C_8 and switch S (in position 1 for f.m. reception), and is also applied to the control grid of the type 6U5 cathode ray tuning indicator tube.



FIG. 21. Cascade limiter circuit employing two limiter tubes. The .a.v.c. voltage is obtained from the first tube. The General Electric Model JFM-90 f.m. converter is one example of f.m. units employing this cascade limiter arrangement.

For a.m. reception, the output of the i.f. transformer for a.m. is fed into the diode detector (a 6H6 tube with both plates tied together), and the a.f. voltage component across R_5 is fed to the a.f. amplifier through d.c. blocking condenser C_2 . The d.c. voltage across R_5 is fed to the a.v.c.-controlled tubes through a.v.c. filter R_7 - C_4 and switch S (in position 2 for a.m. reception), and is also applied to the tuning indicator tube.

Cascade Limiter. Improved limiter operation both for weak and strong signals, along with considerably higher gain, can be secured with two limiter tubes connected as shown in Fig. 21, in what is known as a cascade limiter circuit. The action of the first limiter tube is essentially like that of the limiter circuit in Fig. 19A, with the grid resistor and condenser being chosen to give an even shorter time constant so this first limiter will effectively reduce impulse noise. Whatever amplitude variations exist in the output of the first limiter tube are removed by the second limiter tube, so that the f.m. signal which the limiter section feeds into the discriminator has essentially constant amplitude. Listening tests have shown that this cascade limiter reduces impulse noises far more than any single-tube limiter circuit.

One engineer recommends that the r.m.s. value of limiter output voltage be at least 20 volts for effective noise reduction. The minimum voltage gain for the entire r.f. system including the limiter should be at least 20,000,000 times to permit reception of weak desired f.m. signals without noise interference. To provide this amount of amplification, an r.f. amplifier stage ahead of the frequency converter is highly desirable.

The F.M. Detector

The f.m. detector or discriminator used in an f.m. receiver to convert f.m. signals into audio signals is much like the frequency discriminator circuit employed in the a.f.c. system of a conventional a.m. superheterodyne receiver. Technically speaking, the discriminator converts deviations in frequency from an assigned resting value into positive and negative d.c. voltages corresponding to the original audio signal.

Although the action of a.f.c. discriminator circuits has been covered elsewhere in the course in connection with a.m. receivers, the action will be reviewed briefly here in connection with the typical f.m. discriminator circuit given in Fig. 22.

Discriminator transformer T_1 is of special design, with its primary winding L_2 tuned to the 10.7-mc. i.f. value by C_2 , and with a center-tapped secondary winding L_3 tuned to the same i.f. value by C_3 . The limiter output current flowing through L_2 induces a corresponding f.m. voltage in L_3 , and resonant step-up produces across the two sections of L_3 the voltages e_1 and

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 e_2 , which are always equal in magnitude and are 180° out of phase with each other when considered with respect to center-tap p.



FIG. 22. Typical discriminator circuit for an f.m. receiver. The discriminator load resistors are the 100,000-ohm resistors R_1 and R_2 .

Limiter output voltage $e_{\mathbf{P}}$ in series with e_1 is applied to diode section 1 of the 6H6 double-diode tube. (D.C. blocking condenser C_1 provides an r.f. path from the upper end of L_2 to p, while the chassis and r.f. by-pass condensers C_5 , C_6 and C_7 complete the path for r.f. signals from the lower end of L_2 to the cathode of diode section 1. Thus, $e_{\mathbf{P}}$ exists across choke coil L_1 , and acts in series with e_1 or e_2 on either diode.) Likewise, limiter output voltage $e_{\mathbf{P}}$ in series with e_2 is applied to diode section 2.

The net voltage applied to each diode section is, therefore, the vector sum of the two individual voltages acting on that section. Each diode section rectifies its net applied r.f. voltage and produces a proportional d.c. output voltage across its load resistor.

The load resistor for diode section 1 is R_1 , and the load resistor for diode section 2 is R_2 . The chassis provides the connecting path between the lower end of R_2 and the cathode of diode section 2.

Electrons flow in opposite directions through R_1 and R_2 , as you can readily

verify by tracing the diode circuits. This means that the combined voltage across both R_1 and R_2 , which is the output voltage of the discriminator, will at each instant be the difference between the individual voltages. If the individual voltages are equal, the discriminator output voltage will be zero; if the voltages across R_1 and R_2 are different, the combined voltage will have the polarity of the larger of the two individual voltages, and will be equal in magnitude to their numerical difference.

Let us consider now the factors which make the output voltage of one diode higher than that of the other. First of all, we must choose some voltage or current for reference purposes. Since $e_{\mathbf{P}}$ is common to all circuits under study, we can use it as our reference voltage.

Phase relationship in this discriminator circuit must be considered for three different conditions: 1. When the limiter output signal frequency is equal to the i.f. resting frequency to which the discriminator resonant circuits are tuned; 2. When the limiter output frequency is *less than* the i.f.



FIG. 23. These vector diagrams will help you to understand the action of the discriminator in an f.m. receiver.

resting frequency; 3. When the limiter output frequency is higher than the i.f. resting frequency. The vector diagrams for these three conditions are shown at A, B and C respectively in Fig. 23, with primary voltage $e_{\rm P}$ serving as the reference vector in each case.

The r.f. voltage e_8 which is induced in secondary winding L_3 is 180° out

Under-Chassis Parts

Turning the chassis of our superheterodyne receiver upside down, as in Fig. 15, we find that most of the actual wiring and practically all of the small radio parts are underneath. As a result, the bottom of the radio chassis may look somewhat complicated to you now.

Don't let this worry you, however. A radio man never has to figure out all at once what every single part under the chassis is doing. When he works



FIG. 15. When you turn the chassis of our superheterodyne receiver upside down for a casual inspection, this is what you see. A radio man doesn't care how complicated a set looks underneath its chassis, however, because he looks only for certain parts or circuits and disregards all the rest of the things underneath the chassis.

underneath the chassis, he is looking for one particular circuit or part, and disregards all the other parts.

In this section, you will see a number of under-chassis views, each with only certain groups of parts showing. These views really represent what the trained eyes of a radio man might see as he works on the chassis.

Soldered Joints

Poorly soldered joints between groups of wires or between wires and terminals are a possible cause of receiver failure, but bad joints are just about the easiest defects to repair. There are over 75 soldered connections which can go bad in our superheterodyne receiver. A loose joint can cause annoying crashing sounds to be heard with the radio program from the loudspeaker when the receiver is jarred, or can momentarily affect receiver performance in other ways.

Servicing Technique. Here is how you could go about locating a bad joint. Take the chassis out of the receiver cabinet and set the chassis upside down in such a way that you can see all of the joints under the chassis. Now take a stick of wood and push against each of the joints or leads in turn under the chassis while the receiver is in operation and tuned to a station.

If you hear the crashing sound from the loudspeaker as you move one of the joints or its leads, you know that the joint is loose and should be resoldered. A joint can be loose and defective even though it looks good, because it takes only a very mall separation between wires to ause trouble.

After you learn a bit more about radio fundamentals and servicing techniques, you will become able to tell which section or stage has a defective joint, and will then have to check only the joints in that section.

Resistors

A careful count reveals that there are nine small carbon resistors mounted underneath the chassis of our superheterodyne receiver. The actual-size view in Fig. 16 of one of these resistors is typical of all the rest, for carbon resistors look much alike. The construction of this resistor is shown in Fig. 17.

Each resistor is in a current-carrying circuit. The current flowing through

of phase with the primary r.f. voltage $e_{\rm P}$, hence $e_{\rm S}$ is shown 180° out of phase with reference vector $e_{\rm P}$ in each of the vector diagrams in Fig. 23.

When the limiter output signal is exactly at the i.f. resting value of 10.7 mc. to which the discriminator circuits are tuned (when no sound-is being transmitted), secondary tuned circuit L_{3} - C_{3} is at resonance, and secondary current i_8 flowing through L_8 will be in phase with es, as indicated in Fig. 23A. The voltage produced across the entire secondary winding by this secondary current will therefore lead both is and es by 90°. The resulting. secondary voltage $(e_1 + e_2)$ across L_3 is utilized only in connection with the center tap of L_3 , however, so if we show e1 leading is by 90°, we must show e_2 as lagging e_1 by 180°, just as in Fig. 23A.

Adding $e_{\mathbf{P}}$ and e_2 vectorially gives E_2 as the resultant voltage acting upon diode section 2. Likewise, adding $e_{\mathbf{P}}$ and e_1 vectorially gives E_1 as the resultant voltage acting upon diode section 1. The vector diagram in Fig. 23A shows that these two voltages are equal for the no-modulation condition, and hence the d.c. voltages developed across R_1 and R_2 by the two diode sections are equal in magni-The resultant voltage across tude. both R_1 and R_2 is therefore zero, just as it should be, since no a.f. signal should be obtained when there is no a.f. modulation at the transmitter.

When the limiter output signal frequency is lower than the i.f. resting value to which resonant circuit L_3-C_3 is tuned, this circuit becomes capacitive, and i_8 leads e_8 , as shown in Fig. 23B. Since voltages e_1 and e_2 must be 90° out of phase with i_8 , we have the unequal resultant voltages E_2 and E_1 , as in Fig. 23B. With diode section 1 getting the higher r.f. voltage E_1 , we secure a higher d.c. voltage across R_1 than across R_2 , and the combined

voltage across R_1 and R_2 is therefore positive with respect to ground. The more the limiter output frequency swings below the i.f. resting frequency, the greater will be this positive voltage applied to the a.f. amplifier input.

By a similar analysis, we obtain the vector diagram shown in Fig. 23C for the condition wherein the limiter output frequency is *higher* than the i.f. resting frequency. The net voltage applied to the input of the a.f. amplifier by R_1 and R_2 combined is now negative with respect to ground.

The frequency discriminator circuit shown in Fig. 22 thus produces at its output a d.c. voltage which is at each instant proportional to the deviation in the incoming signal frequency from its resting value, and having a polarity determined by the direction in which this frequency deviation occurs. The discriminator thus converts an f.m. signal directly into the original audio signal voltage used for modulation purposes at the f.m. transmitter.

R.F. by-pass condensers C_5 and C_6 in Fig. 22 must have a low reactance at 10.7 mc., and yet must have a high reactance at audio frequencies so there will be no serious shunting effect upon the a.f. voltage developed across R_1 and R_2 . This a.f. voltage is fed to volume control R_8 through d.c. blocking condenser C_4 , which prevents the C bias voltage of the first a.f. amplifier tube from entering the discriminator circuit and prevents the d.c. discriminator output voltage from acting on the grid of the first a.f. tube.

S Curve for Discriminator Action The solid-line curve in Fig. 24 shows the relationship between the incoming r.f. signal frequency of an f.m. receiver (with respect to the resting value) and the d.c. output voltage of the discriminator. Because of the similarity of this curve to the letter S, it is commonly known as an S curve. This curve, representing the characteristics of the f.m. receiver up to the input of the audio amplifier, should be linear over the entire deviation range of the incoming f.m. signal, for otherwise amplitude distortion would be present in the a.f. output of the discriminator.

The linearity of the S curve in Fig. 24 depends upon two things, the design of the discriminator transformer, and the over-all response characteristic of all the stages ahead of the discriminator. The over-all response is determined chiefly by the dynamic characteristic of the limiter; the response of stages ahead of the limiter affects the limiter only when signals are too weak to cause saturation of the limiter for the entire deviation range.

In designing the discriminator transformer, the Q factor of each resonant circuit and the coupling between the two windings are particularly important: these factors must make the discriminator characteristic combine with the r.f.-limiter characteristic in such a way that the individual resultant voltages, E_1 and E_2 , will at each instant be proportional to the frequency deviation. When this condition is attained, the combined discriminator output voltage across R_1 and R_2 will be proportional to the frequency deviation, thereby giving across these resistors the desired a.f. voltage without amplitude distortion.

It is desirable to have the S curve linear over a somewhat wider range than the maximum deviation, to compensate for inaccuracies in tuning, frequency drift in the local oscillator, or misalignment of the discriminator resonant circuits.

A falling off in the amplitude of the limiter output near the deviation limits, due to too weak a signal at the limiter input, will reduce the frequency range over which the S curve is linear. The dash-dash S curve in Fig. 24 illustrates this condition.

When the S curve for discriminator action has too short a linear region for weak signals, reproduction will still be satisfactory at medium and low program loudness levels, but amplitude distortion will occur during



FIG. 24. S curves for an f.m. receiver, showing how the discriminator output voltage varies with frequency deviation. Curves like this are obtainable only if the over-all grid voltage-plate voltage characteristic of the limiter is flat beyond the point at which saturation begins (like curve 1 in Fig. 18A). When this over-all limiter characteristic is not flat (is like 2 or 3 in Fig. 18A), the angles which the curves make with the horizontal axis will be different (giving either lower or higher discriminator output voltages), and the curves will no longer be linear (resulting in distortion of the audio output voltages). You will often find an S curve like this drawn from the lower left to the upper right. Reversal of connections to the discriminator diode plates gives this result; either connection is satisfactory, for the polarity of an a.f. signal cannot be detected by human ears.

loud sounds. Thus, all sounds loud enough to cause a deviation of more than 40 kc. will swing the signal frequency beyond the linear region of the dash-dash curve in Fig. 24. This is why it is so important that the limiter in an f.m. receiver deliver a constantamplitude output signal over the entire deviation range. The goal of the f.m. receiver designer is to provide sufficient voltage gain ahead of the limiter so that the limiter can maintain constant output amplitude for the weakest desired incoming signal.

Alignment of Limiter and Discriminator Stages. During servicing of an f.m. receiver, the limiter and discriminator can only be aligned for best possible operation. Little can be done to alter the linearity of the discrim-



FIG. 25. This diagram shows how section S of the band-changing switch in an f.m.-a.m. receiver can be used to connect either the discriminator or the a.m. second detector to the input of the audio amplifier. Note that volume control R_0 is effective at both switch settings.

inator, for this linearity is controlled chiefly by the limiter characteristics and the design and construction of the discriminator transformer.

Alignment Procedure. Connect a signal generator to the input of the final i.f. amplifier stage for f.m., set this s.g. to the i.f. resting frequency (usually 10.7 mc.), then open the grid return lead of the limiter circuit, insert a .01-mfd. condenser at this

point, and connect across the condenser a 0-200-microampere d.c. meter.* (In circuit diagrams of f.m. receivers, you will often find in the grid return lead of the limiter stage a condenser, about .01 mfd., shorted out by a lead. This condenser is inserted for measuring the grid current during alignment of the limiter: the service technician simply unsolders the shorting lead, then connects his microammeter across the condenser. This is illustrated in Fig. 16, where C_4 would be the .01mfd. condenser provided for measuring purposes. The microammeter would be connected to points a and b, after first opening the lead between these two points.)

With these preliminary connections made, turn on all apparatus, adjust the limiter input trimmers (C_1 and C_2 , in Fig. 16) for maximum grid current as indicated by the meter, adjust the signal generator output voltage so that the meter reading is at least 50 microamperes, then readjust C_1 and C_2 a final time for a maximum meter reading. This 50-microampere or larger current is necessary so the limiter will provide normal loading on the preceding resonant circuit.

Without changing the s.g. settings, remove the microammeter (or vacuum tube voltmeter) from the limiter input circuit and restore the connection between points a and b. Now connect a high-resistance voltmeter across one of the diode load resistors (across either R_1 or R_2 in Fig. 22), and adjust the discriminator transformer primary trimmer (C_2 in Fig. 22) for maximum output voltage.

Next, connect the high-resistance voltmeter across both R_1 and R_2 (a

^{*} With limiter circuits which also provide an a.v.c. voltage, a vacuum tube voltmeter connected to the a.v.c. source can be used in place of the microammeter. The adjustment would be made for maximum a.v.c. voltage, as this would then correspond to maximum grid current,

connection between the cathodes of the diode sections in Fig. 22 will do this), and adjust the discriminator transformer secondary trimmer (C_3 in Fig. 22) for zero d.c. output voltage. This last adjustment is usually quite critical.

This completes the alignment of the limiter and discriminator stages. The remainder of the tuned circuits in an f.m. receiver are then aligned, usually for peak response, by adjusting each alignment trimmer in turn either for maximum a.v.c. voltage or for maximum grid current in the limiter.

The Audio System. The over-all fidelity of an f.m. receiver is determined to a great extent by its audio system. If the receiver is intended only for f.m. reception with a minimum of noise, and high fidelity is not required, an ordinary a.f. amplifier and loudspeaker will probably be employed. When high-fidelity f.m. reception is desired, however, the audio amplifier will be designed to handle a frequency range from about 30 cycles to about 15,000 cycles, and a specially designed reproducing system will be used to handle this extremely wide range of audio frequencies.

The sound-reproducing system in a high-fidelity f.m. receiver will usually employ at least two loudspeakers, one to handle low and intermediate frequencies, and the other to handle the higher frequencies. These loudspeakers will be coupled to the audio amplifier output stage through special networks, in order to secure proper division of output energy to the two loads. The entire sound-reproducing system will be housed in an acoustically corrected cabinet. The normal rating of the audio system will ordinarily be at least 15 watts, to secure. sufficient power for proper reproduc-, tion of loud bass notes.

In the transmission of f.m. signals, it is customary to accentuate or preemphasize the higher-frequency notes at the transmitter, so that these will over-ride noise. To cut these high frequencies back down to normal, high-fidelity f.m. receivers have a high-frequency attenuator in the audio amplifier. This high-frequency attenuator (usually a simple resistorcondenser circuit) compensates for the pre-emphasis at the transmitter and thus restores the normal balance between highs and lows.

In a combination f.m.-a.m. receiver, a single audio system will serve for both types of reception. A switch like that shown in Fig. 25 will ordinarily be provided to connect the audio amplifier input circuit to the detector stage being employed for a particular program. This switch will be ganged to the main band-changing switch in the r.f. system, so as to make detector switching automatic.

In the circuit of Fig. 25, 50,000-ohm resistor R and .001-mfd. condenser Cin the f.m. channel form the attenuator which counteracts f.m. transmitter pre-emphasis of high audio frequencies. All commercial f.m. transmitters will have the same amount of preemphasis, and this will be such that an attenuator having a time constant of 75 micro-seconds will restore normal receiver response for high audio notes. Normal attenuation of the a.f. amplifier at high audio frequencies. due to stray shunt capacities between circuit leads and ground, must be taken into account when designing the attenuator. Thus, the values used for R and C in Fig. 25 give a time constant of only 50 micro-seconds, but the a.f. amplifier has stray shunt capacities which make the total attenuation equivalent to that of a 75micro-second attenuator.

If no switch were employed at the a.f. amplifier input (if both points 1 and 2 in Fig. 25 were connected to the a.f. amplifier input permanently), the

diode detector not in use would place an additional load on the other diode during positive peaks, causing amplitude distortion of the audio signal. For example, if an f.m. signal were being received, point a in Fig. 25 would be made alternately positive and negative with respect to point b and the chassis by the audio signal. This in turn would make point c in the unused a.m. detector alternately positive and negative with respect to the chassis. Since point c is connected to the plates of the unused diode through resistor R_{3} , the plates would likewise be positive for half of each cycle, and would be conductive. The path through R_8 and the diode would limiter. Comparing the circuit in Fig. 26 to the discriminator, you will find that one diode is reversed, so that we have a half-wave rectifier rather than a push-pull type. Now the voltage across C_2 and C_3 is the sum of the two rather than their difference. Hence, the audio must be taken off in a bridge arrangement from terminals m and n.

When the signal input is such that k is positive with respect to l, the diodes both conduct, causing C_4 to be charged (in parallel with C_2 and C_3 in series) to a value corresponding to the average carrier level of the signal. With no modulation, the voltages e_1 and e_2 , added vectorially to e_3 , are equal, so the C_2 and C_3 drops are



FIG. 26. A ratio detector.

then be a load on the f.m. detector during each positive a.f. peak, thus reducing the positive peaks. To prevent this condition, the detector switch is used.

Ratio Detectors. Recently, there have been introduced several hybrid circuits that perform both the functions of limiting and discrimination in a single tube. One example of this is the "ratio" detector, also known as the Seeley detector. Basically, it is like the discriminator you have been studying, except that it is insensitive to amplitude changes and hence acts as its own equal. Hence *m* is at the same potential as *n*, and there is no audio. When the e_1-e_3 vector is longer than the e_2-e_1 one, the drop across C_2 is greater than that across C_3 , and vice versa.

If we assume that the initial C_4 voltage is 10 volts, then the C_2 and C_3 sum must equal this, or be 5 volts each at the resting frequency. Then, should the signal make the C_2 voltage 8 volts, and the C_3 voltage 2 volts, point mwill be 3 volts more negative than formerly. (Point n remains fixed by the C_4 drop.) When the signal swings the other way, then point m moves positive. Hence, we get our audio output from the frequency swings, as in the discriminator circuit. \vdash

The C_4 - R_1 - R_2 time constant is made purposely long, so that the voltage across it follows the average amplitude of the incoming signal. This fixes the potential of terminal n. Also, should there be any sudden amplitude change, such as may be produced by noise, there can be no quick change in the C_4 voltage; hence the C_2 - C_3 sum remains relatively fixed. The noise pulse is dropped across the diodes because they pass more current trying to charge C_4 to a new value. As long as the C_2 - C_3 sum is constant, only frequency variations can cause an audio output by shifting the C_2 and C_3 volttion at the transmitter, and that the frequency deviation from this resting frequency be at all times proportional to the sound level at that instant.

Two different methods for securing frequency modulation with the required stability and linearity are widely used. One involves the use of an inductance tube and an automatic crystal monitoring control, while the other, suggested by Major Armstrong, starts with a crystal oscillator and utilizes a phase-shifting circuit. Let us analyze each method in turn.

Inductance Tube Transmitter. A block diagram of a commercial f.m. transmitter of the inductance tube



FIG. 27. Block diagram of an f.m. transmitter employing an inductance tube in the modulator stage. The main oscillator stage is of the ordinary L-C type, but its frequency is corrected automatically by the auxiliary crystal oscillator acting through an automatic frequency control system.

ages individually. Hence, this is a combination limiter-discriminator that is popular in the less expensive sets particularly.

Modern F.M. Transmitters

Although the method shown in Fig. 6 for producing frequency modulation of a carrier signal is simple and effective, it lacks frequency stability; an r.f. oscillator shunted by an inductance tube in this manner will drift in frequency with variations in temperature and with variations in the d.c. supply voltage.

In commercial f.m. broadcast stations, it is essential that the transmitter return to its assigned resting frequency whenever there is no modulatype is given in Fig. 27. In this system, the oscillator is of the conventional L-C type, with temperaturecompensated parts in its tank circuit (no crystal). This r.f. oscillator is tuned to some fraction of the assigned resting frequency of the f.m. station. and frequency-multiplying stages are employed ahead of the final amplifier to bring the frequency up to the assigned resting value. Thus, in the f.m. transmitter chosen as an example for Fig. 27, the r.f. oscillator is operated at one-ninth the assigned resting frequency, and its output is fed into two frequency tripler stages which bring the signal up to the correct frequency.

The inductance tube in the modu-

World Radio History

lator stage is connected across the tank circuit of the L-C oscillator, and hence any change in the grid voltage of the inductance tube will change the frequency of the oscillator. Two voltages act upon the grid of this inductance tube; one is the audio signal voltage applied-by the speech input amplifier, and the other is a d.c. voltage produced by the discriminator in the following manner.

Let us assume that the L-C oscillator is at exactly one-ninth of the assigned resting value, and that there is no audio modulation. By means of weak capacitive coupling to some point in the r.f. amplifier, the signal at the assigned resting value is fed into a mixer stage, where it beats with a lower fixed r.f. signal produced by a temperature-controlled crystal oscillator acting through frequency-multiplying stages. The result is a signal at the difference frequency, corresponding to the i.f. signal in a superheterodyne receiver.

The mixer output signal is fed into a discriminator circuit which is tuned exactly to this difference frequency, and which therefore produces a d.c. voltage which is proportional to deviations from this difference frequency. With the L-C oscillator exactly correct, there is negligible deviation, and the d.c. output voltage which the discriminator applies to the grid of the inductance tube is therefore zero.

When the L-C oscillator drifts in frequency, the discriminator produces a d.c. voltage which is proportional to the magnitude of the frequency drift and has a polarity determined by the direction of the drift. This d.c. voltage changes the inductance of the inductance tube exactly enough to correct the frequency drift.

The action of this automatic frequency-correcting circuit is the same when audio modulation is present, for a condenser filter in the discriminator prevents audio signals from traveling from the discriminator to the inductance tube. Only the d.c. voltage actually due to drift from the assigned resting frequency can act upon the inductance tube.

Sometimes a portion of the audio_ output of the discriminator is utilized to provide inverse feed-back which compensates for distortion occurring in the transmitter. In this case, a special coupling circuit would be used to apply both the d.c. and a.f. components of the discriminator output to the inductance tube grid.

The frequency-tripling stage is simply an r.f. stage operated as a class C amplifier, with its plate resonant



FIG. 28. Vector diagrams showing how a frequencymodulated signal can be produced by the Armstrong phase-shift method.

circuit tuned to the third harmonic of the input frequency.

Any tendency for the f.m. transmitter to drift from its assigned resting frequency thus produces a d.c. voltage which acts upon the grid of the inductance tube in such a manner as to correct the tendency toward drift. A zero-center type of voltmeter connected across the discriminator output indicates when the drift is getting dangerously great, thus warning the operator to retune the L-C oscillator manually before the discriminator loses control.

Armstrong Phase-Shift Transmitter. The basic principle underlying the Armstrong method is: When a signal having a definite frequency is flowing through a circuit, and another signal having this same frequency but 90° out of phase is suddenly sent into the circuit, there is an instantaneous phase shift in the original signal, with the result that the frequency of the combination either increases or decreases. We can illustrate this principle by means of the vector diagrams in Fig. 28.

Vector V_1 in Fig. 28A represents the original signal, having a definite frequency and a constant amplitude. As this vector rotates in the conventional counter-clockwise direction at a rate corresponding to its frequency, let us suddenly introduce into the circuit a voltage V_2 which has the same frequency as voltage V_1 but leads V_1 by exactly 90°. The sudden introduction of this new voltage V_2 causes the net voltage V of the combination to shift in a counter-clockwise direction, corresponding to a momentary increase in frequency. If voltage V_2 is suddenly increased in amplitude after it is introduced into the circuit, resultant vector V speeds up again, producing another momentary increase in frequency.

On the other hand, if the suddenlyintroduced V_2 lags V_1 by 90°, the resultant vector V will move clockwise, corresponding to a momentary decrease in frequency, as indicated by the vector diagram in Fig. 27B. If vector V_2 is varying continually in both amplitude and polarity in accordance with a sine-wave audio signal, resultant voltage vector V will swing above and below its original frequency in a corresponding manner.

All this means that we can produce frequency modulation of a carrier signal by combining it with another signal which has the same carrier frequency but is alternately 90° leading and 90° lagging the original signal in accordance with positive and negative swings of an audio signal, and varies in amplitude in accordance with variations in the amplitude of the audio signal. In order to make the deviation in frequency dependent only upon the *amplitude* and *polarity* of V_2 , an attenuator is introduced into the audio amplifier of the transmitter so as to make the audio output voltage decrease gradually as the audio frequency increases. This insures that a 1-volt audio signal at 10,000 cycles will produce the same frequency deviation as a 1-volt audio signal at 1000 or 100 cycles.

As a rule, both V_1 and V_2 in an f.m. transmitter of this type will have relatively low frequency values, and fre-



FIG. 29. Basic circuit used in the Armstrong phase-shift type of f.m. transmitter.

quency multipliers will be used to bring the f.m. signal up to the correct frequency for the transmitting antenna.

The basic Armstrong circuit for securing frequency modulation by means of this phase shift method is given in Fig. 29. Vacuum tube VT_1 in a crystal oscillator circuit produces the initial low r.f. carrier value (200 kc. would be a typical value) with negligible frequency drift. The output of this crystal oscillator is fed into r.f. amplifier tube VT_4 and also into the balanced r.f. stage including tubes VT_2 and VT_8 .

Note that the grids of this balanced stage are fed *in phase*, while the output of this stage goes to an untuned r.f. transformer having a centertapped primary L_2 - L_3 . With in-phase plate currents flowing in opposite directions through the primary to the center tap, and with the tubes balanced so that the currents in the two sections of the primary are normally equal in magnitude, the resultant flux linked with secondary winding L_4 is zero, and no voltage is induced in this winding for transfer to the frequency multipliers through amplifier stage VT_{5} .

The values of the parts in the plate circuits of the two balanced amplifier tubes are such that the r.f. current in each half of the primary lags the a.c. grid voltage by approximately 90°; in other words, the plate loads for VT_2 and VT_3 are essentially pure inductances.

The operation of this circuit depends upon the basic fact that when the a.c. resistance of a tube is varied. its a.c. plate current will vary correspondingly. We can change the a.c. plate resistance of a tube by varying the screen grid voltage.

In the circuit of Fig. 29, the screen grid voltages for both VT_2 and VT_3 in the balanced amplifier stage are applied through the secondary winding of audio transformer T. With this arrangement, an a.f. voltage applied to the primary winding will make the screen grid voltage on one tube increase, and make the screen grid voltage on the other tube decrease. This causes an unbalance in the r.f. plate currents flowing through L_2 and L_3 , and consequently we secure a resultant flux which links with L_4 .

resultant current flowing The through both L_2 and L_3 will either lag or lead the a.c. grid voltage by 90°, depending upon which coil current is greatest. This means that the r.f. voltage induced in secondary winding L_4 will either lead or lag the a.c. grid voltage by 90°, depending upon whether the a.f. input voltage is swinging positive or negative, and the amplitude of this r.f. voltage in L_4 will vary in accordance with the amplitude of the a.f. input signal. The voltage across L_4 thus corresponds to vector V_2 in the diagrams of Fig. 28.

Amplifier stage VT_5 merely provides a means for securing the proper relationship between the amplitudes of r.f. voltages V_1 and V_2 . These two voltages, from tubes VT_4 and VT_5 , are fed in parallel into the frequency multiplier system. They combine to give the desired frequency-modulated signal, as was explained in connection with the vector diagrams in Fig. 28.

#8 Riproduction will be satisfactory for meduin & low programs Tortion toudness but amplitude distortion will occur during loud sounds. 36

TEST OUESTIONS

Be sure to number your Answer Sheet 34FR-2.

Place your Student Number on every Answer Sheet.

Send in your answers for this lesson immediately after you finish them. Doing this insures that the graded answers will reach you while the subject matter is still fresh in your mind, and you will get the greatest possible benefit from our speedy personal grading service. Never hold up a set of lesson answers.

- 1. To secure a high signal-to-noise ratio with minimum inter-station interference at receivers, should the maximum frequency deviation in an f.m. system be high or bear
- 2. In what two ways can an interfering signal affect a desired f.m. signal? FREQUENCE & ANIPLITUDE
- 3. What is the purpose of the discriminator in an f.m. receiver? 4. What section in an f.m. receiver produces the a.v.c. voltage?
- LIMITER SECTION
- 5. What section in an f.m. receiver removes amplitude variations from the f.m. signal? LIMITER
- 6. What characteristic should a limiter have in order to block out sudden. It must be fast acting having a short time constant
- 7. Why is it permissible to have a peaked over-all response for the r.f. and i.f. sections in a properly designed f.m. receiver? of the force day The limiter will flatten the overall support actions over the in 8. Describe the type of distortion which occurs when the S curve for dissections over the ente
- criminator action has too short a linear region for weak signals.
- 9. Why is it essential that grid current be drawn by the limiter tube during alignment of the resonant circuit ahead of the limiter?

Solis Uniter will provide normal location, on the 10. Why is an attenuator for high audio frequencies used in the audio system of an f.m. receiver? at the transmitter and rectories

the normal balance between

histor louis.

a resistor produces a voltage drop across it. This voltage drop controls voltage and current in various ways both in the d.c. supply circuits for tubes and in the circuits which transfer signals from tube to tube.

Resistor Color Code. Carbon resistors are by far the commonest of all resistors used in radio equipment. Each resistor is marked with rings of color which tell the electrical size in



FIG. 17. Inside story of a carbon resistor. The two wire leads (pronounced LEEDS) make contact with the ends of the resistance material.

ohms just as accurately as would printed values. Elsewhere in the course, you will be shown how to read the color rings on resistors.

Resistor Troubles. A resistor can fail because of a break in one of its leads, usually inside the resistor or at the point where the wire lead enters the body of the resistor. A crosswise crack in the carbon resistance element can also cause failure of a resistor. In both cases, the radio man says that the resistor is *open*, for the break or crack blocks electron flow through the resistor.

Locating a Bad Resistor. A defective resistor can always be located by making tests with servicing instruments, chiefly with an ohnumeter. **Burned-Out Resistors.** Resistors also become defective due to overheating, which occurs when an excessively large current flows through the resistor.

Overheating may cause the resistor to open up or change its electrical value, but radio men will simply say that the resistor has burned out. Such a resistor can usually be spotted at a glance because its colors and markings are blackened by the heat. A blackened resistor is not necessarily defective, however; it may have been overloaded at some time, but not enough to cause failure.

Condensers

Four different types of condensers are to be found underneath the chassis of our superheterodyne receiver paper condensers, electrolytic condensers, mica condensers, and trimmer condensers. The illustration in Fig. 18 shows all of these condensers, with other parts painted out for clearness.

In basic electrical action, all four types of condensers are alike, in that they all control voltage and current in signal and supply circuits. The chief differences in condensers occur in the method of construction and in the electrical sizes.

Paper Condensers. The condenser used more than any other type in radio equipment is the *paper condenser*. This name comes from the fact that the material used for insulation inside the condenser is waxed *paper*. The construction is shown in Fig. 19.

Loose Leads. A poor connection between a foil strip and one of the wire leads is a common cause of trouble in a paper condenser. A defect of this nature can sometimes be located by wiggling the body of each condenser under the chassis while the receiver is turned on. The defective con-

SHALL I CHANGE JOBS?

Before you can answer this question, you must set your goal in life. Do you want a professional career as Radiotrician or radio operator? An executive position in the radio industry? A radio business of your own?

With a goal in mind, apply these three questions to your present job: 1. Does it give reasonable pay for the present? 2. Does it give knowledge, training or experience which will be worth money to you in the future? 3. Does it give you prestige, or bring you in touch with men who can help you to attain your goal?

Judge each new job opportunity also by these three questions. Hold each job only long enough to learn what is needed for the next job. Then, if a vacancy up ahead is unlikely, you are justified in changing to a corresponding job somewhere else.

The important thing is to keep going ahead. Learn something every day. When you reach the point where you are no longer learning, no longer progressing ahead, then you are already moving backward, and it's time for you to change jobs.

First, though, be sure you really are prepared for a better job. Be sure your training is complete, and be sure you are able to apply it.

J. E. SMITH





Study Schedule No. 35

For each study step, read the assigned pages first at your usual speed, then reread slowly one or more times. Finish with one quick reading to fix the important facts firmly in your mind, then answer the Lesson Questions specified for that step. Study each other step in this same way.

 I. Introduction; Automatic Tuning Systems; Problems of Automatic Tuning; Mechanical Automatic Tuning Systems - - - Pages 1-9 This lesson, which completes the foundation of your Radio career, contains a wealth of thoroughly practical information on push-button tuning systems. Only fundamental facts are given—basic knowledge which will enable you to master in a jiffy the intricacies of any specific system you may encounter in your work. Answer Lesson Questions 1, 2 and 3.

☐ 2. Electrical Automatic Tuning Systems - - - - - Pages 9-14 The first time you work on a set employing this system of tuning, you'll see the importance of the fundamental facts in this section. Pay paricular attention to the material dealing with adjustment of trimmer condensers. Answer Lesson Question 4.

☐ 3. Electro-Mechanical Automatic Tuning Systems - - - Pages 14-18 The various types of motor-driven tuning systems are classified here in such a way that you can readily recognize each type in an actual receiver and quickly figure out how it should work. Some of the most fascinating mechanisms in Radio are described here. Answer Lesson Questions 5 and 6.

4 Audio-Silencing and A.F.C.-Releasing Switches for Electro-me-

chanical Systems - - - - Pages 18-23 When you tune a receiver manually, one station after another blasts from the loudspeaker. When a motor does the job, however, a simple switch keeps the set silent until the desired station is tuned in. You learn how this is done, and also study two complete electro-mechanical systems which are presented as examples. Answer Lesson Question 7.

□ 5. Tuning Motors - - - - - - Pages 23-28 Although you will invariably replace defective tuning motors rather than attempt to repair them, a knowledge of how they work is interesting and useful background information for a Radio man. Pay particular attention to the methods used for reversing these motors, because trouble often develops in the reversing switches. Answer Lesson Questions 8, 9 and 10.

6. Mail your Answers for this Lesson to N.R.I. for Grading.

☐ 7. Typical Receiver Diagrams and How to Analyze Them - - Reference Text 35X

Study carefully now the explanation of the a.c.-battery portable receiver on pages 1-6, so as to familiarize yourself with these recently-developed sets which operate from a single self-contained storage battery.

Plan to study the other circuits in this text along with the first five lessons of your Advanced Course, one circuit per lesson.

8. Start Studying the Next Lesson.

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Automatic Tuning Control Systems

Introduction

A UTOMATIC tuning, which permits tuning in a desired station accurately and more or less instantly simply by pressing a lever or button, is rapidly reaching the point where it is being considered essential in a re-The average listener soon ceiver. learns that he can obtain practically all desired programs from a few nearby broadcasting stations, with a clarity and freedom from interference which cannot be obtained in the case of distant or foreign stations. This listener then wants to be able to change quickly from one station to another without having to bother with accurate tuning, so he can choose the most interesting program. In making these things possible, automatic tuning has fulfilled a real need among radio listeners.

Automatic Tuning Systems

Although manufacturers are using many different schemes for providing automatic tuning, we can divide these into three groups according to the operating principle employed, as follows:

1. Mechanical Automatic Tuning Systems. By pressing a button or rotating a telephonetype dial, the listener himself provides the force required to rotate the gang tuning condenser to the setting for a desired station. This is a purely mechanical action, with no electrical switching whatsoever; tuning is essentially instantaneous.

2. Electrical Automatic Tuning Systems. Pressing a button switches an entirely new set of condensers, pre-adjusted to a particular station, into the tuning circuit of the receiver in place of the gang tuning condenser. The action here is entirely electrical, hence tuning is instantaneous.

3. Electro-Mechanical Automatic Tuning Systems. Pressing a button closes the circuit to a small electric motor, which then rotates the gang tuning condenser to a desired station. Electrical switching here causes a mechanical force to be applied to the gang tuning condenser. A certain amount of time is required, once a button is pressed, for the motor to complete the tuning process.

In all three systems, the initial adjustments which insure accurate automatic tuning to desired stations must be made by the radio dealer at the time of the installation. Printed tabs having the call letters of the desired stations are attached to the pushbuttons themselves or to the escutcheon surrounding the buttons, to identify the station selected by each button.

Problems of Automatic Tuning

Proper design and installation of an automatic tuning device does not insure accurate automatic tuning indefinitely; many other factors must be considered in the tuning device itself and in the circuits which are controlled.

Mechanical Vibration. All three automatic tuning systems are subject to vibrations and jars, especially at the time the buttons are pressed, which may alter the initial adjustments. The inaccuracies of tuning introduced by vibration may not be noticeable when local stations are tuned in on a receiver having a reasonably broad response, but when automatic tuning is employed on a highly selective receiver for the reception of semi-local or distant stations we really have a serious problem. Automatic frequency control (A.F.C.) is an effective solution, though somewhat expensive; improvements in the mechanical design of the automatic tuning system can greatly reduce undesirable effects of vibration, making tuning sufficiently accurate without A.F.C. provided that automatic tuning is restricted to local stations.

Frequency Drift. Changes in the output frequency of the local oscillator due to changes in temperature present another serious automatic tuning system problem, particularly when A.F.C. is not used. Oscillator frequency drift is most noticeable in the high-frequency bands of an allwave receiver, but there can be appreciable frequency drift in the broadcast band as well. In a conventional manually tuned receiver, this frequency drift can be compensated for by shifting the tuning dial setting slightly after the receiver has warmed up and reached a constant internal temperature, but with automatic tuning systems the listener has no means of correcting for this drift.

Investigation has revealed that the chief cause of oscillator frequency drift in an ordinary superheterodyne receiver is the fact that a rise in temperature will increase the dielectric constants of insulating materials which are present in the oscillator circuit." Some dielectric materials are affected more than others by changes in temperature; when materials such as molded plastics, bakelite, low-grade rubber insulation and waxed paper insulation are present in the oscillator circuit, the result is an increase in the effective capacity in the oscillator tuning circuit as temperature goes up, producing a reduction in the oscillator output frequency

and consequently lowering the I.F.

By using only dielectric materials like mica and ceramics (which are little affected by temperature) for oscillator circuit parts, engineers have succeeded in reducing frequency drift. In addition. special temperaturecompensating condensers are often shunted across the regular condensers in the oscillator tuning circuit. These condensers are each made up of a fixed metal plate and a movable plate to which is connected a bi-metallic strip. Air forms the dielectric between the two plates of this compensating condenser. Increases in temperature cause corresponding changes in the shape of the bimetallic strip, and these in turn cause the movable plate to move farther away from the fixed plate, reducing the capacity. The amount and nature of compensation required to offset frequency drift due to temperature changes varies with each receiver. design.

Muting. In some mechanical and electro-mechanical tuning systems the receiver is tuncd quite slowly through many stations before the desired one is reached, creating an annoying situation if the receiver is not silenced in some way during the actual tuning process. It is for this reason that the audio system is often silenced or muted during the time the tuning mechanism is being rotated.

Eliminating A.F.C. While Tuning. An A.F.C. circuit tends to hold on to stations while passing through them, so it is perfectly possible for a receiver to "hang on" to a station adjacent to that corresponding to a particular push button. For this reason a switch is built into the automatic tuning mechanism to disconnect the A.F.C. system temporarily while the station is being tuned in.

^{*} As you will recall from earlier lessons, the dielectric constant of a material is equal to the capacity of a condenser with that material between its plates divided by the capacity of the same condenser when only air is between the plates. The higher the dielectric constant of the material used in a condenser, the higher will be its capacity.

Mechanical Automatic Tuning Systems

Mechanical automatic tuning systems may be divided into two general groups according to the manner in which they are operated by the listener:

1. Rotary or telephone dial types, in which the listener himself provides the rotary motion which turns the tuning mechanism to the correct setting for a desired station. Automatic stops prevent him from moving beyond the correct setting.

2. Direct push types, in which the listener applies a direct push or force to a button or lever. Either a gear, cam or lever arrangement is used to convert this force into the rotary motion required to turn the tuning condenser to the correct setting for a desired station.

Telephone Dial Type. The basic principles involved in a telephone dial type of mechanical automatic tuning system are illustrated in Fig. 1. An analysis of the problems involved in this purely imaginary system will make it easier for you to understand the more complicated systems actually used.

In Fig. 1A the rotor and stator of a conventional tuning condenser are shown in the minimum-capacity or highest-frequency position. As the rotor meshes more and more with the stator plates, the capacity increases and the receiver tunes to a lower frequency. A circular metal disc, on the rear side of which are several metal pins or pegs, is attached to the shaft of this tuning condenser. Suppose that a 1300 kc. station can be tuned in by rotating the rotor clockwise through the angle marked θ_1 ; if the pin on this circular metal disc makes contact with a stop stud the instant this tuning condenser setting is reached, we can readily see that this 1300 kc. station will be tuned in automatically.

Now suppose that turning the tun-

ing condenser rotor through an angle θ_2 tunes in a 1100 kc. signal. A pin placed at the position marked 1100 on this circular disc will automatically stop the rotor at the correct position for this station. As many additional pins as are desired can be placed upon the circular metal disc, one for each station.

At once we see a mechanical difficulty. The pin for the 1300 kc. station will prevent the rotor from turn-



IG. 1. Basic principles of the telephone-dial type mechanical automatic tuning system.

ing to the positions required for lower-frequency signals. Suppose, however, that the pin is made in the manner shown in Fig. 1B, with a spring which normally holds the pin outward so it can pass over the stop stud. Pressing in the button and rotating the entire mechanism clockwise causes this pin to strike the stop stud, stopping the rotor at exactly the correct position for a desired station. Still another mechanical difficulty arises. With the construction shown in Fig. 1A, the positions of the pins cannot be changed and consequently the choice of stations is limited to those selected at the time of construction. We could eliminate this objection by mounting the pins in a circular slot like that shown in Fig. 1C, so they could be moved to the correct position for a desired station. This arrangement prevents selecting adjacent stations, however (each pushpin is made in the form of a crank arm which can be rotated to any desired position when the button assembly with its gear teeth is pulled out of the correspondingly grooved sleeve. In this way the effective end portion of the pin can be set at any position within the swing area of this crank arm, eliminating entirely the need for slots in the circular metal disc.

To silence or mute the receiver when a button is pressed and the mechanism rotated, the pin could well



FIG. 2. Constructional details of the Emerson automatic dial mechanism as used in models AT-170. AT-181 and other Emerson receivers. When a station is tuned in, the button selected for that station will be near the top of the dial. One of the eleven holes shown at A is not used, and is covered by a blank space on the bakelite cover plate. When operating this dial mechanism, the direction in which the dial is rotated must be such that the blank space does not have to pass across the top of the dial, for the rotor plates are then either completely meshed or completely unmeshed, and cannot be rotated farther. Since this dial rotates through 360°, and ordinary tuning condenses rotate only through 180°, it is obvious that a gear drive is used between the dial shaft and the condenser shaft.

button has a certain minimum thickness), but we could overcome this difficulty with a staggered arrangement of slots, as illustrated in Fig. 1D.

Obviously, a staggered slot arrangement would not give a pleasing arrangement of push-buttons on the receiver dial. A symmetrical arrangement can be secured, however, by utilizing the construction shown in Fig. 1E; here the position of the button is fixed on the metal disc, and the be made to depress a flexible metal ring as it moves, causing this ring to contact a fixed metal ring behind and thus short the input of an audio amplifier stage, as illustrated in Fig. 1F. Releasing of the push-button after the station setting has been reached would open the audio circuit muting switch.

Emerson Automatic Dial. A mechanical automatic tuning system of the telephone dial type is used on a number of Emerson receivers. The operating principle is essentially the same as that illustrated in Fig. $1E.\sqrt{}$

Constructional details of this Emerson automatic dial mechanism are shown in Fig. 2A. A number of pushbuttons, each mounted on a pin which is shaped like a crank arm (see Fig. 2B), are arranged symmetrically on the circular dial. A station call letter tab, protected by a transparent celluloid cap, fits into a circular recess in the top of each button. The , genious pivoted or floating vane serves to stop the tuning condenser assembly at exactly the same point regardless of the direction in which the dial mechanism is rotated. A spring normally holds the vane in the middle position shown in Fig. 2C, and serves also to absorb the shock when the button pin is suddenly forced against the vane. Fixed stops, shown on each side of the vane in Figs. 2Dand 2E, prevent the vane from swing-





FIG. 3. Three views of an automatic tuning system used on a number of Philco receivers which also have A.F.C. The numbered parts on these diagrams are all identified in the Philco service bulletin on automatic tuning; the important parts are also identified in this lesson.

sides of each button are grooved in much the same manner as a gear, as' also are the holes in which the buttons slide; this procedure prevents a button.from turning in its hole yet allows it to be moved in or out.

To operate this Emerson automatic tuning dial, the button corresponding to the desired station is pressed and the entire dial then rotated until the pin hits a stop which prevents further rotation; this tunes in the desired station correctly. In this unit an ining beyond a definitely limited angle.

Since this particular mechanism is used chiefly on inexpensive receivers which do not employ A.F.C., no provisions are made for blocking the audio system or shorting the A.F.C. system during the tuning process.

Philco Automatic Tuning Dial. The general appearance of the Philco telephone type automatic tuning dial with cover plate removed is shown in Fig. 3A. In using this unit, the lever arm is swung to the position corresponding to the call letters of the desired station, the knob on the end of the lever is pushed in (engaging the lever with the tuning dial) and the entire mechanism is rotated in either a clockwise or counter-clockwise direction (only one direction will be correct) until a stop is reached. A click will be heard an instant before the knob reaches its final position, signifying that a spring-operated



FIG. 4A. Operating principles of direct push types of mechanical automatic tuning systems employing a cam with either a lever and roller or a plunger. Pressure on the button in the direction indicated by the heavy arrow serves to rotate the gang tuning condenser (geared to the cam shaft) to the correct setting for the station assigned to that button.

switch has temporarily shorted the A.F.C. system in order to allow the desired station to "take hold." \bigvee

Constructional details of this mechanism can be seen by studying the front-view diagram of Fig. 3B and the cross-section view in Fig. 3C. Pushing in the knob at the upper or outer end of the lever 29 does three things: 1, it causes a metal pin on this lever to set itself in one of the holes marked 28 (Fig. 3B); 2, it depresses the key corresponding to the desired station (one of these keys is shown as item 15 in Fig. 3C): 3, it causes insulated switch blade 16 to make contact with shaft bushing 19. thus shorting the audio system dury ing the tuning process.

With the lever button depressed, the entire dial mechanism is rotated until the depressed key engages with the stop assembly (x in Fig. 3B). Note the long slit which locks the key pin. On this assembly, item y is a pivoted flapper; movement of the pin over this closes switch contacts \mathcal{S} in Fig. \mathcal{SC} , shorting the A.F.C. system temporarily. When the pin falls into the recess between the two flappers, this switch opens, and the A.F.C. system then makes the final correction in tuning. Releasing the pressure on the knob and lever opens the audio shorting switch, making the program audible.

In setting up this Philco automatic dial tuning system, a station is first



Courtesy A. W. Franklin Mfg. Corp.

FIG. 4B. Franklin direct push type automatic tuning unit, an example of the mechanism shown in dotted lines at the left in Fig. 4A. Note that all cams are on the same cam shaft, with a separate locking screw holding each cam in its correct position for a particular station. The cams used here have the same shape as that shown in Fig. 4A, except that the unused part of each cam (the shaded portion in Fig. 4A) is cut away.

tuned in the conventional manner, \vee with the A.F.C. or magnetic tuning control in its "out" position. The key (item 15) at the bottom of the dial is then adjusted by inserting a screw driver in its slot, pressing the key in slightly so it is free to rotate, then turning the key until a click is heard. The receiver is now tuned for maximum output by turning the key back and forth slightly. This procedure is repeated for each other station selected.

Direct Push Types of Mechanical Automatic Tuning Systems. A number of different mechanisms are being used to convert an ordinary direct push on a button into rotation of a tuning condenser to the correct setting for a desired station; let us look over a few of them.

First of all, keep in mind that a gang tuning condenser must be rotated through an angle of 180° (one-half of a complete revolution) to cover an entire tuning band. Because of the mechanical difficulties involved in converting a direct push into a full 180° of rotation, it is customary to use an auxiliary shaft which can be rotated through a maximum angle of 90° or less, with the gears between this shaft and the tuning condenser shaft to step up this rotation to the required amount.

Heart-Shaped Cam with Lever and Roller. In receivers employing this



FIG. 5. Egg-shaped cam and plunger mechanism used in the mechanical automatic tuning systems of some Philco receivers.

particular tuning unit, there will be one complete set of parts like those in Fig. 4A for each station which is to be tuned automatically. A downward pressure on the station-selecting button is applied to the heart-shaped cam through a roller at the other end of the lever arm, forcing the cam to rotate to a position which brings the roller to point G, closest to the center of the shaft. For example, with the cam in the position shown, pressure on the button will cause the cam and shaft to rotate clockwise. The cams for the different stations are mounted side by side on a cam shaft which is geared to the tuning condenser shaft, h





FIG. 6B (below). Chassis of Philco model 39-17 superheterodyne receiver, with automatic tuning mechanism visible at the right of the gang tuning condenser. Bakelite buttons are pushed over the ends of the six plungers after the chassis is mounted in its cabinet. Note the gear drive between the cam shaft and the tuning condenser shaft. The tuning control knob is placed on the cam shaft; the locking screw which holds all cams rigidly in position (by friction with spacing washers) fits into the center of this knob.

the cams being separated by spacing washers and held in position by friction. A mechanical locking device is provided for locking each cam rigidly in position once it is adjusted for a station.



FIG. 18. If only the condensers underneath the chassis of our superhet receiver were visible, this is what you would see. The condensers not marked are paper condensers.

denser will usually cause a change in loudspeaker volume when this is done.

Shorted or Leaky Condensers. If the waxed paper insulation between the two foil strips of a paper condenser fails completely (becomes punctured, so that the metal strips touch each other through a small hole in the paper), the condenser is said to be shorted.

Just as the high voltage between the points of an auto spark plug can burn a hole through a sheet of paper, so can high voltage burn a hole through the insulation of a paper condenser. This paper insulation weakens with age, so that eventually the normal circuit voltage is sufficient to short the condenser.

If the insulation becomes partly defective, so that a few electrons can flow from one strip to the other through the paper, the condenser is said to be *leaky*. Simple measurements with an *ohmmeter* will locate shorted or leaky paper condensers.



FIG. 19. This photo shows you what's inside a paper condenser. Two long strips of metal foil separated by strips of waxed paper are rolled together, with the two foil strips projecting from opposite ends of the roll and making contact with the two wire leads. The cardboard housing and fiber end discs keep out moisture.

Cam with Plunger. In some systems a straight plunger, with or without a roller, is used in place of the lever, as indicated by the dotted lines at the left in Fig. 4A. The push-button tuner shown in Fig. 4B is an example of this particular mechanism; a %-inch movement of the button here provides a 60° maximum rotation of the cam shaft, and this can be stepped up to 180° with a 3-to-1 gear ratio. tween the cam shaft and the tuning condenser shaft is clearly visible in this photo. \checkmark

Finger and Rocker. A finger and rocker mechanism which provides mechanical tuning in still another manner is illustrated in Figs. 7Aand 7B, "a sketch of this unit" is shown in Fig. 7C. For each station there is a plunger (flat metal strip) sliding freely through two slots in opposite sides of a metal frame. At



FIG. 7. The diagrams at A and B show one of the finger and rocker units used in the mechanical automatic tuning systems of some Crosley receivers (including the Crosley Safety-Tune auto radio), while the sketch at C shows the complete tuning unit with a gear drive to a gang tuning condenser. Tightening the screw on the plunger locks the finger rigidly in position. To set up a button for a station, this screw is loosened so the finger can rotate, the button is pushed all the way in, the station is tuned in manually, and the screw is tightened to lock the finger at the correct angle. A spring returns the button to its normal position when pressure is released, leaving the tuning condenser at the correct setting for the desired station.

Egg-Shaped Cam with Forked Plunger. In another system of the direct push type, illustrated in Fig. 5A, the cam is somewhat egg-shaped and the roller is replaced by a Ushaped or forked metal piece. Pressing the button makes the forked plunger take the position shown in Fig. 5B, holding the cam in a definite position. Photographs of the Philco version of this arrangement appear in Figs. 6A and 6B. The gear drive beone end of this plunger is the pushbutton; clamped to one face of the plunger is a metal "finger" which can be set at any desired angle to the plunger and held in position by a locking screw and clamp arrangement (omitted from Figs. 7A and 7B to simplify the diagrams, but shown in Fig. 7C). Pressing in a button makes the rocker rotate to the same angle as the finger; on the rocker is a gear segment which meshes with a gear on


FIG. 8A. Sprague electrical automatic tuning unit with built-in trimmer condensers. The switch blades in the foreground are normally held apart by the strip of insulating material mounted on each plunger; when a button is pressed, the metal inset moves down between the blades, shorting them and closing the circuit to one set of trimmer condensers.

the tuning condenser shaft and thus provides the correct tuning condenser setting for the station assigned to that button.

In many of the receivers which incorporate a mechanical automatic tuning system, some form of compensating condenser is employed in the oscillator circuit to compensate for temperature changes which might otherwise alter the accuracy of the automatic tuning system. The tuning action is so rapid that audio silencing switches are usually omitted.

Electrical Automatic Tuning Systems

Instead of rotating the tuning condenser when a new station is desired,





Courtesy Spraque Products Co.

FIG. 8B. Bottom view of the Sprague unit, showing the trimmer condensers and their adjusting screws. Each push-button controls one upper and one lower trimmer on this gang assembly. One terminal of each trimmer is grounded to the frame of the unit, and this frame is in turn grounded to the receiver chassis.

electric automatic tuning actually removes the variable condensers in the tuned circuits and replaces them with new condensers which were previously adjusted to the correct values for that particular desired station.

Push-button switching mechanisms like that shown in Fig. 8A are used in electrical automatic tuning systems. When one of the buttons on this unit is pressed down, the button which formerly was down is released, removing that set of condensers, and an entirely new set of condensers is switched in. The entire process of switching is practically instantaneous. It is common practice to mount the set of pre-adjusted condensers right on the switching mechanism. In Fig.





World Radio Histor

8B is a bottom view of the unit in Fig. 8A; as you can see, there are two trimmer condensers, each adjusted by a screw, for each of the buttons on the unit.

Now let us see exactly what happens when one of the push-buttons in the unit shown in Fig. 8 is pressed. The diagram in Fig. 9A gives a side view of the button assembly in its out and open position, while that in Fig. 9B shows the same assembly with the button pushed in. First of all, observe that the button is on the end of a sheet metal plunger which slides freely in and out through slots in the frame of the switching unit. A spring



tuned circuit of the receiver. The fact that one terminal of each tuning condenser section in a receiver is grounded makes it possible to use this simple switching system. ν

A separate push-button and plunger assembly like that shown in Fig. 9Ais required for each station which is to be automatically tuned, but only one latch bar is needed for the complete switching mechanism. Pressing of any button lifts up the latch bar; this serves to release any other plunger which may have been previously pushed in and held by the latch bar. Since stations can be tuned in just as fast as the buttons can be



Courtesy P. R. Mallory & Co., Inc.

FIG. 10. Photo and cross-section diagram of Mallory-Yaxley multiple push-button switch. Trimmer condensers are not built into this unit, but can be mounted either on the sides or at the rear of the unit; each trimmer must then be connected to one of the soldering lugs on the insulated terminal board. This unit is available with many different combinations of contacts and terminals, and can be used to switch in three trimmer condensers at a time in receivers having three tuned circuits.

under compression, acting between the plunger and the left-hand frame in Fig. 9A, normally holds the plunger entirely out. When the button is pushed in, a pivoted flapper or latch bar falls into a groove in the plunger, as indicated in Fig. 9B, preventing the plunger from returning to its normal out position.

A strip of insulating material, with a U-shaped metal piece set into each end, is attached to the plunger. Pressing in the button causes each metal piece to move between a pair of switch contacts, shorting the contacts and thus inserting a set of pre-adjusted trimmer condensers in the pressed, there is no need for audio silencing switches or A.F.C. releasing switches.

Although many different forms of switching mechanisms are being used for condenser substitution purposes in electrical automatic tuning systems, all have essentially the same construction as that shown in Fig. 9. There will be some form of plunger for each push-button, with switching contacts or blades mounted on the plunger but usually insulated from it. Likewise there will always be a latch bar to hold a plunger in when a button is pressed. A photograph of one other switching mechanism employing these principles, together with diagrams showing its construction, are given in Fig. 10.

Typical Circuits. In Fig. 11 is shown the circuit for the preselector and oscillator sections of a typical superheterodyne receiver employing two tuned sections. Tube VT is a conventional pentagrid converter. Condenser $C_{\mathbf{P}}$ of the gang tuning condenser tunes the preselector coil L_2 to resonance, while C_0 tunes the oscillator coil L_3 . Only three connections. need be made to this circuit to incorporate electrical automatic tuning; one lead from the push-button switching mechanism goes to point x, which is the stator terminal for preselector tuning condenser $C_{\mathbf{P}}$. Another lead goes to point y, which is the stator terminal of the oscillator tuning condenser C_0 , and the third lead is grounded to the tuning condenser frame or to the chassis.

Semi-Automatic Connection Observe that the gang tuning condenser section in Fig. 11 is still present in the circuit after the push-button pushing mechanism is connected: this is known as a semi-automatic connection, because the gang tuning condenser must be turned to its minimum-capacity (highest frequency) setting before automatic tuning can be used. When this is done, pressing of any one of the four station buttons serves to tune in a station. For example, pressing button 1 places trimmer condenser C_{P1} across the preselector tuned circuit and places trimmer condenser C_{01} across the oscillator tuned circuit.

Since an automatic tuning system provides for reception of only a limited number of stations, it is obvious that some means for restoring manual tuning must be incorporated in the receiver. (Manual tuning cannot be used when the automatic tuning system is connected, because of the fact that one set of trimmer condensers will always be in shunt with the gang tuning condenser section.) \checkmark An extra button is often provided on a push-button tuning for the sole purpose of restoring manual tuning. This button is designated as M in Fig. 11, and serves to open the two ungrounded leads to the push-button switching mechanism. Pressing button M, which is often marked



FIG. 11. This circuit shows a semi-automatic connection of an electrical automatic tuning unit to a superheterodyne receiver employing a pentagrid converter tube. Pressing button *M* restores manual tuning; pressing any other button inserts a set of preadjusted trimmer condensers (ona trimmer for the oscillator tuning circuit and one for the preselector) in the tuning circuits of the receiver, after first breaking previous trimmer connections. Each set of trimmers is adjusted initially to exact resonance for one desired station.

MANUAL, opens switches S_P and S_O and at the same time releases any other button which may have been pressed in.

Full-Automatic Connection. Naturally, it is somewhat of a nuisance to turn the tuning dial to its highest frequency setting whenever automatic tuning is desired. This undesirable characteristic of the semi-automatic connection can be eliminated by modifying the switching mechanism so it will disconnect both sections of the gang tuning condenser from the receiver circuit whenever automatic tuning is used. Fig. 12A illustrates one way in which this can be accomplished, and Fig. 12B shows how the MANUAL button could provide the required double-pole, double-throw switch action.

Oftentimes it is undesirable to have long leads running from the receiver chassis to the push-button tuning mechanism (because of the inductance of these leads); for this reason



FIG. 12. At A is a simplified circuit showing full-automatic connection of an electrical automatic tuning unit to a superheterodyne receiver having two tuned stages. Pressing button M restores the two tuned stages. Pressing button M restores the gang tuning condenser connections and breaks the two connections to the tuning unit; pressing any other button releases M, disconnecting the gang tuning condenser sections, and inserts a set of trimmers in the tuned circuits. The sketch at Btrimmers in the tuned circuits. The sketch at B shows how this switching could be accomplished with a plunger like that in Fig. 9.

the manual-tuning switch is sometimes incorporated in the band-changing switch. In this case one setting of the switch might be labeled BROADCAST MANUAL and another setting labeled BROADCAST AUTOMATIC. In one receiver the change-over from manual to automatic operation is accomplished by pushing the wave band switch knob in or out; this particular switch also operates dial lights which illuminate the words automatic or manual to indicate which particular tuning system is being used.

Tuning by Coil Substitution. Just as with mechanical automatic tuning systems, the oscillator tuned circuit can be the cause of frequency drift in an electrical automatic tuning system. To secure better frequency stability, some manufacturers are using a pushbutton switching system to substitute adjustable coils instead of trimmer condensers in the oscillator tuned circuit. Special coils employing pulverized iron cores which can be moved by means of an adjusting screw to change the inductance of the coil are used for this purpose. A fixed condenser, usually of the temperaturecompensating type, provides the necessary capacity for the oscillator circuit. Because of the higher cost of variable-permeability iron core coils, they are usually used only in the oscillator circuit. A slight change in trimmer condenser capacity will have far more de-tuning effect in the oscillator tuned circuit than in a preselec-

tor tuned circuit.

Initial Adjustments. It is neither advisable nor necessary to make each adjustable part in an electrical automatic tuning system cover the entire 540 to 1500 kc, broadcast band, A more economical and stable construction is secured by limiting the tuning range of each adjustable coil or trimmer condenser to a definite section of the broadcast band; for example, one set of adjustable parts may be designed to tune from 540 to 900 kc., another set may cover the range from 700 to 1300 kc., and the third and final set might cover the range from 1000 to 1500 kc. Notice that there is enough overlapping between these three groups so that a station near the limit of one group may also be tuned in by another group.

Receivers employing electrical automatic tuning are usually adjusted by the servicemen as a part of the

installation job, rather than by the listener. First of all, it is wise to prepare a list of the stations which have been selected by the listener for automatic reception. Assign one pushbutton to each of these stations, making sure that each button is in the proper frequency group, as specified in the instruction manual supplied with the set. The station call letter tabs, also supplied by the manufacturer, can now be affixed to the buttons or the escutcheon. The receiver should be turned on during this preliminary work, so it can warm up for twenty to thirty minutes and reach its stable operating temperature. If you suspect that the I.F. amplifier may be out of alignment, be sure to check this and realign if necessary before making any preselector and oscillator adjustments.

To set up a button for a station, tune in the station manually and note the nature of its program at that time. Now push in the button assigned to that station (if a semi-automatic connection is used, be sure to turn the tuning condenser to its minimum-capacity position before adjusting any trimmer condensers); locate the oscillator trimmer condenser or variable inductance controlled by that button, and adjust until the desired station is heard with maximum audio output. For best results do not depend upon your ears, but use an output indicator or the tuning indicator in the receiver (if available). With this done, locate the preselector trimmer condenser which is controlled by this button and adjust for maximum output in the same manner; you will note that this adjustment is quite broad, whereas the setting of the oscillator trimmer was quite critical. Repeat this prosedure for each other push-button.

Radiotricians sometimes prefer to

use a signal generator which is set to the station frequency and fed into the receiver input as a guide for initial adjustments of the trimmer condensers; if the signal generator is quite accurately calibrated, only slight additional adjustments of the trimmers will be needed after the signal generator is removed.

Oftentimes the initial adjustments will change slightly during the first



FIG. 13. Circuit diagram of the tuned sections of the RCA-Victor model HF-1 high-fidelity broadcast band superheterodyne receiver, designed for automatic reception of eight different local stations. Eight push-buttons control the electrical automatic tuning system; there is no tuning dial or knob.

few days after they have been made. When the receiver does not have A.F.C., it is a good idea to return in about a week and make final adjustments which will correct for any changes which may have occurred.

RCA-VICTOR Model HF-1. This is an excellent example of a receiver employing a push-button switching mechanism for substituting a pre-adjusted variable inductance in the oscillator tuned circuit and substituting a pre-adjusted trimmer condenser in each of the two preselector tuned circuits. No manual tuning control is provided. The circuit diagram for the tuned sections of this receiver is given in Fig. 13; vacuum tube VT is a conventional pentagrid converter tube, with the first two grids acting as the oscillator grid and plate respectively. The oscillator tuned circuit is made up of condenser C, fixed inductance L, and one of the adjustable-core oscillator coils shown just below in the diagram. Coil L acting alone with condenser C resonates below the correct frequency, and the pre-adjusted coil which is shunted across L by the switching mechanism lowers the net inductance in the circuit exactly the correct amount for a particular station. Coil L_6 serves as the oscillator feed-back coil.

Two tuned circuits in the preselector make up a band pass tuner. In one of these circuits is fixed inductance L_2 and one of the pre-adjusted antenna trimmer condensers which is controlled by the push-button switching mechanism; the other preselector tuned circuit is made up of coil L_3 and one of the trimmer condensers. All three input circuit coils, L_1 , L_2 and L_3 , are mutually coupled inductively.

Provisions are made for automatic tuning of eight different stations. The switching action is as follows: Pressing button 1 closes switch S_{01} , shunting L_{01} across oscillator coil L, and simultaneously closes switches S_{A1} and S_{P1} , inserting pre-adjusted trimmer condensers C_{A1} and C_{P1} in their respective preselector tuned circuits. Pushing in button 5 would first release all switches controlled by button 1, then close switches S_{05} , S_{A5} and S_{P5} , which would place parts L_{05} , C_{A5} and C_{P5} in the tuned circuit.

Electro-Mechanical Automatic Tuning Systems

Electro-mechanical automatic tuning systems can be divided into two general types according to the method of operation involved: 1, the nonhoming system; 2, the self-homing system. Each type will generally include the following sections:

1. A small electric motor which drives the gang tuning condenser through speed-reducing gears and which can be reversed by means of a switch.

2. A switching mechanism which can be adjusted to stop the driving motor at predetermined positions which correspond to the gang tuning condenser settings for desired stations.

3. A group of push-button-controlled switches, each of which starts the motor and connects into the motor circuit the proper switch mechanism for stopping the motor at the correct point (these may be located at any reasonable distance away from the receiver, making remote control tuning possible).

4. A means for silencing the audio system of the receiver during the interval when the motor is driving the tuning condenser, in order to prevent annoying blasts of sound as the receiver is tuned past strong undesired stations; a means for releasing temporarily the A.F.C. system while the tuning motor is in motion or just after it stops, in order to allow the desired station to "take hold" of the A.F.C. system.

In the non-homing system the motor always drives the variable condenser in the same direction in which it was rotated by the previous automatic tuning action. If this direction is toward the desired station, the motor stops when it has driven the tuning condenser to its correct setting. If the direction is away from the desired station, the motor is automatically reversed when the tuning condenser reaches the limit of its rotation, and the condenser is then driven past its original setting to the desired new setting. In this non-homing system it is important that the motor stop instantly when its circuit is

opened by the switching mechanism, for otherwise it might coast long enough while power is off to drive the tuning condenser beyond the correct setting. This is called a non-homing system because the motor does not always "go home" to the correct new setting immediately.

In the self-homing system, the electric motor always travels in the correct direction to bring the tuning condenser to the desired new setting in the quickest possible time; if the Variable condensers can be made which will rotate through a full revolution, but it then becomes difficult and costly to make the rotor plates sufficiently rigid.

Non-Homing Electro-Mechanical Tuning Systems. The essential features of a non-homing system are illustrated in Fig. 14. First of all, observe that a small reversible electric motor drives the shaft of the main tuning condenser through a chain of speed-reducing gears. These gears



FIG. 14. Essential features of a non-homing electro-mechanical automatic tuning system.

motor overruns the proper stopping position because of inertia after power has been cut off, it will instantly reverse and correct its error automatically.

In both the self-homing and nonhoming systems the variable tuning condenser cannot swing through more than 180° (one-half of a complete revolution), because the rotor plates of the tuning condensers used in modern receivers are given added rigidity by a metal strip at one end. are necessary because the midget motors used in these systems rotate at a speed higher than one thousand revolutions per minute, a speed which is far too great for tuning purposes.

For simplicity we will consider a system having only two station-selecting buttons, designed as x and y in Fig. 14. For each button there is one metal disc on the extended tuning condenser shaft, held rigidly in place by a friction clamp. A small block of insulating material is set into the circumference of each disc, as indicated. Above each disc is a contact spring which normally makes contact with the disc but which is insulated from the disc whenever the insulating block is directly under the spring. Each contact spring is connected to the switch controlled by-its-push-button; the push-button switching mechanism is the same as that used in electrical automatic tuning systems, with a latch bar serving to release a previously pushed button before holding in the button for a desired station.

Now let us see what happens in this non-homing electro-mechanical system when button x is pushed in. We see immediately that the switch controlled by this button shorts together points 1 and 15, so let us start at point 1 and trace the circuit through the motor to point 15 again. Follow through from point 1 to point 2, then through the contact spring to point 3 and the metal disc provided for this button. From the disc we trace through the tuning condenser shaft from point 4 to point 5 and then through a shaft contact spring to points 6 and 7. From point 7 we know that current must flow through the motor and come out either at point O or point 8. Tracing from point O, we find an open circuit at point 18 and know that current cannot flow over this path at this time. We therefore trace from point 8 to point 9 and then to point 10, the other terminal of the reversing switch. From 10 we go through the metal arm of the reversing switch to point 11, to point 12, to point 13, through a secondary winding on the power transformer (which provides the required voltage for the motor) to point 14, through manual tuning switch M, and finally we are back at points 15 and 1 again.

With this complete circuit for the

motor between its terminals and the source of power, the motor begins to rotate, and the direction will be such that the tuning condenser shaft and the metal disc will rotate in a counterclockwise direction (as indicated by the arrow on the metal disc) until the insulating block in this disc comes directly under the contact spring. This breaks the circuit between points 2 and 3, opening the motor circuit and stopping the motor. If the position of this metal disc on the condenser shaft is properly chosen, the tuning condenser setting will now be exactly correct for receiving the station assigned to button x.

Now suppose that we desire to receive the station assigned to button y. We push in this button, and the latch bar lifts just enough to release button x and open the circuit between points 1 and 15. The latch bar then holds in button y, and the corresponding switch shorts points 16 and 17. Since the arm at the left end of the condenser shaft has not yet touched the reversing switch, contacts 9 and 10 are still together and the circuit traces from point 17 through the lefthand metal disc, through the shaft, through the shaft contact spring to point 7, through the motor to point 8, through contacts 9 and 10 and the reversing switch to point 11, and then through the power transformer winding and switch M to points 16 and 17. The motor rotates in the same direction as it did for button x, and the insulating block on this metal disc moves away from its contact spring. The motor continues rotating in this counter-clockwise direction until the arm at the end of the shaft flips the reversing switch over to a position which opens the circuit between points 9 and 10 and closes the circuit between points 18 and 19. Now the cir-. cuit through the motor is from point 7 to point O and the internal connec-, the circumference of this disc. tions of the motor are such that this reverses the direction of rotation.making the tuning condenser rotate in a clockwise direction. This continues until the insulating block has moved under the contact spring and opened the motor circuit.

The speed of the motor is so high that this tuning action takes place in a few seconds. During the tuning process, the station-selecting knob on the panel of the receiver is rotating and the dial pointer is moving, since both are driven by the tuning condensershaft. (For simplicity, the manual tuning mechanism has been omitted from the diagram in Fig. 14.)VSelf - Homing Electro - Mechanical

Automatic Tuning Systems. The basic features of a self-homing system are shown in Fig. 15. First of all, we note that there is no separate reversing switch for the motor; an ingenious switching mechanism automatically starts the motor in the correct direction, - eliminating the need for reversing the motor during the tuning process.

In this mechanism, a single circular disc of insulating material is permanently and rigidly mounted on the tuning condenser shaft. To the edges of this disc are fastened two semicircular metal segments, with their ends separated by insulating segments J and K so there is no electrical circuit between them. The lead marked 6 connects metal segment v to one metal slip ring which is mounted on but insulated from the tuning condenser shaft, and the lead marked 9 connects the other metal segment (w)to the other metal slip ring. Surrounding and resting upon the two semi-circular metal segments are a number of sliding contacts or brushes, one for each push-button, which can be set at any required position along $\sqrt{}$ which is shorted to point 1 by the

Now let us push in button x and trace through the motor circuit to see what will happen. The switch controlled by button x has only two terminals, 11 and 1, which are shorted together when the button is held in by the latch bar. Starting at terminal 1. we trace to point \mathcal{Z} and through the manual tuning switch M, through power transformer winding S_{M} and then to terminal 3 on the motor. If terminal 4 is the other motor terminal in the circuit, we get a clockwise rotation of the tuning condenser shaft, and if terminal O is in the circuit we



get a counter-clockwise rotation of the shaft. Let us trace from point 4 first: we go to slip ring contact 5, then through the slip ring and through connecting lead 6 to metal segment von the disc. None of the brushes which are resting on this disc provides a closed circuit to point 1 again, so we know that motor current does not take this path.

Returning to the motor, we trace from point O to slip ring contact 8 and through this slip ring and connecting lead 9 to metal segment w. Brush 10, resting on this metal segment, connects directly to point 11,

Capacity. Since one entire lesson of your N.R.I. Course is devoted to condensers of various types, the manner in which condensers are rated according to electrical size and working voltage need not be taken up in this lesson. Now we just want to make a general acquaintance with the different kinds of condensers, so let's pass on to the next type.

Electrolytic Condensers. Two of the condensers in our receiver are of a special type known as *electrolytic* condensers or simply electrolytics. These condensers look much the same as paper condensers, and are made in a similar manner except that there is no paper between the rolled-up aluminum sheets. Instead, the sheets are separated by a moist chemical paste which produces an insulating film on the aluminum sheets through electrolytic action. (A scientist would explain that electrolytic action occurs when an electric current is sent through chemical materials.)

Leaky Electrolytics. Electrolytic condensers are common causes of radio receiver trouble, for the pasty chemical material spoils with age, reducing the effectiveness of the insulating film. Radio men say that the condenser has then become *leaky*, because the insulating film allows electrons to *leak through* the condenser when they should be held back.

Connections. The large electrolytic condenser in this receiver is really two electrolytic condensers combined in one housing. Three insulated leads, colored red, black and blue respectively, come out of one end of the cardboard housing.

The leads are colored for two reasons, to distinguish between the two sections of the condenser and to indicate the correct polarity of connections. An electrolytic condenser will work as a condenser only when connected in a certain way. Correct connections are designated by calling one lead *positive* and the other *negative*; the positive lead must go to the positive terminal of the circuit. In our large electrolytic condenser, the black wire is the negative lead for both sections of the condenser. The red wire is the positive lead for one condenser section, and the blue wire is the positive lead for the other section. This information is marked on the condenser housing.

It is common practice to combine two or more electrolytic condensers in one housing in this way. A careful radio man always makes a rough sketch of connections before removing an old condenser, to reduce chances for errors when connecting the new unit.

Mica Condensers. There is only one mica condenser in our receiver. It is a tiny unit encased in molded brown bakelite, and located just about in the exact center underneath the chassis. (See Fig. 18.) The mica condenser gets its name from the fact that sheet mica is used as the insulation between the metal plates. The construction is shown in Fig. 20.

Possible Troubles. Mica condensers seldom fail in use, but are more costly than paper condensers. Occasionally a wire lead may break at the point where it enters the bakelite body of a mica condenser. The break may not always be visible, but can be located by wiggling each of the mica condensers in turn while the receiver is in operation. When a defective condenser is moved in this manner, it affects the loudness of the sound coming from the loudspeaker.

The mica insulation can fail due to moisture or to excessively high voltages. Adjacent metal plates then switch, and hence we have a closed circuit for the motor. The motor rotates the shaft in a counter-clockwise direction until brush 10 is directly over insulating segment J. This breaks the circuit, and the motor stops. If the motor coasts far enough_ after the circuit is broken so that brush 10 touches metal segment v, the motor circuit is instantly closed through its other path (4, 5, 6, 10, and11 to 1) and the motor starts up in the reverse direction rotating the shaft in a clockwise direction. This quickly returns brush 10 to its position above the insulating segment, and since the motor has not had a chance to build up speed, it stops instantly. If brush 10 is properly located, a station will now be tuned/ in correctly.

Now suppose we wanted to receive the station assigned to button y. We press in this button, releasing button x and shorting points 12 and 13. Current now flows over the path 12-2-3-4-5-6-7-13-12, and the motor turns the tuning condenser shaft in a clockwise direction until brush 7 is directly over insulating segment J. If the motor "over-runs" the insulating segment, brush 7 will contact segment wand the motor will be reversed just long enough to correct the error. In this way a self-homing or self-correcting automatic tuning action is obtained.

Although only two station-selecting buttons have been shown in Figs. 14 and 15, as many additional buttons can be used in both these systems as are desired. In each case, one additional manual tuning button M must be pressed to open the main motor supply lead when manual tuning is desired. Since the common latch bar acts upon all buttons, the mere act of pressing a station button releases the manual tuning button, closing its switch and restoring automatic tuning. 🕨

With non-homing systems, one metal disc with its insulating block is required for each station which is to be automatically tuned; with self-• homing systems only one large disc is required;-with-one brush resting on the metal segments of this disc for each station which is to be automatically tuned. The metal segments may be made wide enough so that two brushes can rest side by side or very close together on it, permitting automatic tuning of stations which are only 10 kc. apart.

Audio-Silencing and A.F.C.-Releasing Switches for Electro-Mechanical Systems

There are five methods in general use for silencing the audio system, of which the first is the most common:

1. Grounding the grid of an audio tube.

2. Shorting the terminals of the loudspeaker voice coil or shorting the primary winding of the output transformer.

3. Shorting the diode load resistor in the second detector circuit.

4. Applying a large negative C bias to an audio tube in order to block the tube.

5. Applying a large negative C bias to the tubes in the frequency converter and the I.F. stages of a superheterodyne in order to reduce the sensitivity of the R.F. system.

Two methods are in general use for removing A.F.C. action while the tuning mechanism is in operation or for temporarily releasing the A.F.C. system when tuning is completed:

1. Shorting the output of the A.F.C. discriminator sections.

2. Applying a large negative C bias to the tubes in the frequency converter and I.F. stages. (Reducing the input voltage to the A.F.C. system in this manner prevents the system from holding a station which is even a small amount off resonance.) This method also provides audio silencing.

Regardless of which method is used for silencing the audio system and releasing A.F.C. action, the switches used must be interlocked with the automatic tuning mechanism. The switches can be actuated either by mechanical movement of some part of the tuning system or by the flow of current through the motor circuit.

Switch Control by Motor Shaft Movement. An electric motor consists essentially of a rotating part, called the armature or rotor, and a stationary part called the stator. The interaction of the magnetic fields produced by these two parts causes rotation. For greatest rotating force, the rotor must be exactly in the center receiver operation and A.F.C. action.

When voltage is applied to the motor, the rotor instantly centers itself by moving to the right, as indicated in Fig. 16B. This causes a collar on the shaft of the motor to press against a lever arm which actuates a multi-contact spring switch, shorting all contacts to ground and thus grounding the audio grid and shorting the discriminator output resistor. When current through the motor is interrupted, the rotor-centering force disappears and the spring again pushes the rotor to the position



FIG. 16. Details of audio-silencing and A.F.C.-releasing switches which are actuated by thrust (end-wise movement) of the motor shaft when voltage is applied to the motor of an electro-mechanical automatic tuning system.

of the stator; if the rotor is off center to one end or the other, it will be magnetically pulled into the center whenever voltage is applied to the motor. This endwise movement of the rotor is used in some electromechanical automatic tuning systems to actuate the audio silencing and A.F.C. releasing switches.

The arrangements of parts for a switch control action of this type is shown in Fig. 16A. A spring on the rotor shaft, pressing between the rotor and the frame of the motor, serves to push the rotor off center when the motor is stopped. Under this condition the contacts of the switches are open, permitting normal shown in Fig. 16A, opening the audio and A.F.C. circuits.

The end movement of the rotor shaft is also used to engage the motor with the speed-reducing gears which drive the gang tuning condenser shaft. The movement of the rotor shaft to the right when voltage is first applied causes a drive pin on the motor shaft to engage with a crank arm on the shaft of the smaller gear. as indicated in Fig. 16B. When current to the motor is interrupted by the station-selecting system, the drive pin moves away from the crank arm, allowing the motor to coast to a stop without causing undesirable rotation of the gang tuning condenser. The friction load on the gang tuning condenser itself is sufficient to make it stop instantly when the driving force is removed. The switches shown in Fig. 16 may, of course, be used for any of the other audio silencing and



FIG. 17. Method of using a relay in the motor circuit to actuate audio-silencing and A.F.C.-releasing switches in an electro-mechanical automatic tuning system.

A.F.C. releasing methods which are employed by receiver designers.

Switch Control with Electro-Magnetic Relays. The flow of current in the main lead to the motor of an electro-mechanical automatic tuning of any station-selecting button sends current through the relay coil as well as the motor. The relay armature is attracted to the relay core by the magnetic flux created by the relay coil, and this movement of the armature causes the audio grid and A.F.C. load contacts to be grounded. When the station-selecting mechanism interrupts the current to the motor, the current flow through the relay likewise stops, allowing the spring to move the armature away from the core and open the audio and A.F.C.

Electrical Circuit (without Switches) for Audio Silencing and A.F.C. Releasing. In this circuit, the A.C. voltage for the A.C. motor is also applied to a twin rectifier stage, and the resulting D.C. voltages are made to increase the negative C bias on an audio amplifier tube enough to silence the audio system, and to increase the negative C bias on the



FIG. 18. Circuit diagram showing connections for an electrical system which silences the audio amplifier and releases the A.F.C. system during the operation of the electro-mechanical automatic tuning system. This scheme is used in some Motorola receivers. The use of a duo-diode-pentode in the first I.F. stage eliminates the need for an extra muting tube.

system can be used to actuate an ordinary A.C. relay having contacts which are connected to silence the audio system and release the A.F.C. system, since this motor current flows only during the tuning operation. The circuit arrangement is quite simple and is shown in Fig. 17; the pressing I.F. stages in order to prevent the A.F.C. system from acting during the tuning operation.

A typical circuit arrangement is shown in Fig. 18; the A.C. voltage developed by an auxiliary winding on the motor core is applied to the two diode plates of the first I.F. tube

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through D.C. blocking condensers C_4 and C_3 , the return path being from the grounded cathode to the grounded . end of the auxiliary winding. On alternate half cycles the diode sections act as shorts across resistors R_5 and R_6 , and the current pulses flowing through these resistors on the remaining half cycles produce rectified voltage drops across them.

The D.C. voltage across R_6 acts in series with the normal C bias on the I.F. tube grids. Condenser C_6 places the motor circuit at R.F. ground potential.

RCA Electro - Mechanical Automatic Tuning System. This electromechanical system is of the nonhoming type, and consequently the motor will sometimes drive the gang tuning condenser away from a desired setting until the limit of condenser rotation is reached and a reversing switch is tripped. A top view



FIG. 19. RCA electro-mechanical automatic tuning system of the non-homing type. Shaft coupling units are designated as BB.

the first A.F. tube, driving this tube sufficiently negative to prevent audio amplification. Condenser C_1 and resistor R_7 make up a filter which smooths out ripples in the rectified D.C. voltage across R_6 .

The D.C. voltage developed across R_5 when the motor is operating acts upon the grid of the I.F. tube through resistors R_4 and R_1 , and drives this grid sufficiently negative to reduce the strength of the I.F. carrier signal and thus release the A.F.C. system. Note that this A.F.C releasing voltage acts in shunt with the AV.C. voltage and the normal C bias voltage for the I.F. tube. Condensers C_2 and C_5 are filter condensers, while resistors R_2 , R_3 and R_4 serve to isolate the various negative voltage sources acting upon of the tuning mechanism is shown in Fig 19A; the motor, the speed-reducing gears, the A.F.C. and A.F. shorting switches, and the pin and crank arm drive are all essentially the same as for the system shown in Fig. 16, while the reversing switch is like that shown in Fig. 14.

The simplified diagram in Fig. 19B shows how this system operates. The metal disc with an insulated segment and a slot on opposite sides is one of eight discs which are mounted on a common shaft and held in position by friction; beneath each disc is a spring contact. During manual tuning the metal discs turn with the gang tuning condenser but the electric motor remains motionless since the pin on its shaft is not engaged with

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the crank arm on the speed-reducing gear.

To make the preliminary stationsetting adjustment for this RCA mechanism, one of the buttons is pressed. and after the motor has stopped, an adjusting key (provided with the unit, and kept in a special adjusting key receptacle when not in use) is inserted in the adjusting hole corresponding to this button; this places the key in the slot on the metal disc, exactly as shown in Fig. 19B. The A.F.C. system is turned off by means of a switch on the receiver panel, and the receiver is now tuned manually to the station desired for that button. Removal of the adjusting key comshorting switches are closed. The motor stops when the metal disc has turned to the position where the contact spring is over the insulated segment; if the gang tuning condenser reaches the limit of its travel before this occurs, the reversing switch changes the direction of motor rotation.

General Electric Electro-Mechanical Automatic Tuning System. Diagrams for this non-homing system are given in Fig. 20. As you can see, no metal station-selecting discs are used; instead, a metal arm with a knifeedge spring contact is mounted on the tuning condenser shaft, and swings over a number of metal contact but-



pletes the adjustment for this station; the process is then repeated for each other button. It is not necessary to lock the metal discs in position, since there is sufficient friction to prevent them from slipping during normal receiver operation.

The operation of this automatic tuning system is as follow: The pressing of the push-button station selector in Fig. 19B closes the circuit to the motor through the reversing switch and through the contact spring and metal disc; the motor starts up in the direction of its previous rotation, with its rotor automatically centering in the stator so that the drive pin engages with the crank arm on the gear and the A.F.C. and A.F. tons which are set into slots in an insulating panel. The tuning motor drives the tuning condenser shaft through reducing gears; when the limit of rotation is reached, a reversing switch changes the motor direction and starts the contact arm moving back over the contacts again. All this is shown in Fig. 20A.

The circuit connections for one contact button are indicated in Fig. 20B; here is how the system works. When the MANUAL BUTTON is pressed, it opens the motor supply circuit and raises the latch bar enough to release all other push buttons. The latch bar also releases the OFF BUTTON at this time, thus closing the primary circuit of the power transformer. Under this condition neither the relay nor the motor gets power (because all station selector button switches as well as the manual button switch are open), and the receiver can be tuned manually.

When the STATION SELECTOR BUTTON is pressed, the latch bar releases the MANUAL BUTTON, closing its circuit. Power is thus applied to the motor, and it rotates the gang tuning condenser and the rotating contact arm until this arm is directly over the contact button corresponding to the station selector button which has been pressed. This closes the relay circuit, and its armature pulls in, opening the motor cir-The cuit and stopping the motor. relay remains closed until the pressing of another station selector button switch opens its circuit. The A.F. system is silenced and the A.F.C. system is temporarily released by means of a conventional switching arrangement operated by the endthrust action of the motor (this is not shown in Fig. 20).

Another interesting feature of this General Electric system is the nonlatching SCAN BUTTON. When manual tuning is in use, pushing of the scan button will close the motor circuit, and it will continue to rotate the gang tuning condenser back and forth through its entire 180° of rotation as long as the button is pressed. In this way stations can be tuned in roughly without the trouble of turning a manual tuning knob, and when a desired station is heard the scan button can be released. Accurate final tuning can then be done manually if necessary, but usually the A.F.C. system will be able to take hold instantly and tune in the desired station when the scan button is released.

To set up any station selector but-

ton, the desired station is tuned in manually while the manual button is depressed. A contact button near the contact arm is now loosened and moved directly under this arm. Labeling the station selector button with the call letters of the station tuned in completes the setting up process; this is repeated for each other button.

Tuning Motors

Despite their small size, tuning motors require careful design and construction. They must operate



FIG. 21. Details of a series-wound tuning motor. Short arrows indicate direction of current flow for one half of cycle. Plus sign in wire indicates current flow into paper; dot in wire indicates current flow out of paper.

with as little noise as possible, must operate with little or no oiling, must not heat up excessively while in use, and must operate from an A.C. source. Usually they are designed for 6.3 volt operation, in which case they can be connected in shunt with the filament supply of the receiver; some motors are designed for higher voltages, however, and for these an extra secondary winding must be provided on the power transformer.

There are two types of small A.C. motors in general use in automatic

tuning systems: 1, series motors; 2, induction motors. The fundamental operating principles of each will be discussed.

Series Motors. In Fig. 21 is a simplified diagram illustrating the operating principles of a series motor. -For convenience, the armature is shown as a single loop of wire whose ends are connected to two semi-circular commutator segments; in an actual motor, there will be many loops of wire and many more commutator segments, and the entire armature will be mounted in the center of the stator, between the pole pieces. Two fixed brushes make contact with the commutator segments as indicated. The armature loop and the field coils are connected in series across the A.C. source. as you can easily see by tracing the current from one source terminal through the circuit as indicated by arrows to the other source terminal; all motors having the armature and field coils connected in series in this way are known as series motors. In an actual series motor the wire loop is wound around a cylindrical soft iron core whose shaft is free to revolve in bearings at each end. The stator (the stationary part of the motor) is usually made from sheet metal laminations to reduce eddy current losses. Field coils are wound over the poles of the stator, and are connected together in such a way that both will produce flux in the same direction at any instant. When current flows through this motor in the direction indicated by arrows, the loop of wire on the armature will set up a magnetic flux in the direction indicated, and the field coils will likewise set up a magnetic flux which makes one pole north and the other pole south in polarity, as indicated.

Naturally the magnetic fields produced by the armature and stator will

interact with each other when the armature is inside the stator. For the armature position and direction of current flow shown in Fig. 21B, the armature will act as a pivoted magnet having the indicated polarity. The armature N pole will be repelled by the stator N pole and attracted by the stator S pole, and likewise the armature S pole will be repelled by the stator S pole and attracted by the stator N pole, causing counterclockwise rotation. Since the like poles are closest together now in Fig. 21B, the repelling action will be strongest; when the armature has rotated a bit farther, the unlike poles will be closer together and the attractive force will be the strongest, reach-



FIG. 22. Simplified diagram showing the operating principles of an induction motor.

ing a maximum when the armature N pole is next to the stator S pole. At this time the current through the armature loop is reversed in direction by the commutator, and the entire process of repulsion and attraction repeats itself, causing the armature to continue rotating in the same direction until it is again horizontal and current flow is again reversed by the commutator.

Reversing a Series Motor. With a given connection of armature and field coils, as indicated in Fig. 21A, the armature will revolve in a certain direction; to reverse this direction of rotation it is necessary to reverse the direction of current flow through <u>either the field coil or the armature</u> (but not through both). This can be done by inserting a double-pole. double-throw switch in the circuit as shown in Fig. 21C. If, however, a center-tap field coil arrangement like that shown in Fig. 21D is used, a single-pole, double-throw switch will serve for reversing purposes. Note that with this arrangement only one field coil is carrying current and producing magnetic flux at any one time.

A series motor operates on either an A.C. or D.C. source, since the reversal of current flow through both the field coils and the armature does not change the direction or magnitude of the forces acting on the armature. *Induction Motors.* All induction motors operate on the principle that when a closed or shorted loop of wire is pivoted in the center of a rotating magnetic field, the loop will rotate in the same direction as the field. Let us see why this statement is true.

Suppose we have a magnetic field like that shown in Fig. 22A, made up of an N pole and an S pole, which is being rotated by some mechanical means in the counter-clockwise direction indicated. Between these two poles is a shorted copper loop or ring which is pivoted at its center and is therefore free to turn. If the initial positions of the loop and magnetic field are as shown in Fig. 22A, a certain amount of magnetic flux will pass through the loop. When the magnetic field begins to rotate counter-clockwise, however, the amount of flux through the ring will be reduced and consequently the flux linkage in this coil or loop will be reduced. From Lenz's law we know that whenever there is a change in flux linkage in a coil, a voltage is induced in the coil in such a direction as to oppose the original change in flux linkage. This means that the loop in Fig. 22A will set up a flux to aid the original flux, and the coil

will act as a small magnet having the . polarity indicated. Since like magnetic poles repel and unlike poles attract, we can readily see that the coil will rotate in the same direction as the magnetic field, as indicated by the arrows.

The shorted loop cannot rotate at exactly the same speed as the magnetic field, for then there would be no change in flux through the loop. The loop is continually slipping or lagging behind the magnetic field, but even when it slips past the pole pieces it is still forced in the same direction of rotation. Suppose that the relative positions of the coil and magnetic field



FIG. 25. These diagrams show how a concensor can be made to produce a clockwise-rotating magnetic field for a split-phase induction motor. Short arrows indicate current flow.

are as indicated in Fig. 22B at one instant, with the field traveling faster than the loop. The number of flux linkages in the loop will now be increasing, and consequently the loop will set up a flux to oppose the original flux, in an attempt to maintain a constant number of flux linkages. The polarity of the loop will be as indicated, and since like poles repel we still have the loop rotating in the direction of the magnetic field. In a practical induction motor there are a large number of shorted loops mounted on the armature, each contributing a rotational force.

Electric motor design engineers

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have devised a number of simple ways for producing a rotating magnetic field by electrical means in a single-phase A.C. induction motor (the induction motors used for radio tuning purposes will operate on a two-wire, single-phase, A.C. power_ line, and hence are often called singlephase motors).



FIG. 24. Circuit of General Electric condenser splitphase induction motor as used in a number of automatic tuning systems. When switch SW is set at contact 1, current to poles P₈ and P₄ (which are connected together in parallel rather than in series) must flow through C; when SW is at 2, current to P₁ and P₈ flows through C, reversing the direction of the rotating magnetic field and thus reversing the direction of motor rotation.

Capacitor Split-Phase Induction Motor. In Fig. 23 is shown a method of producing a rotating magnetic field by means of a four-pole stator, with a coil wound on each pole and a condenser connected in series with two opposite coils to make them draw a current which is approximately 90° out of phase with the current through the other coils.

A.C. voltage V, applied to the motor terminals, sends a current ithrough field coils L_1 and L_2 , and the flux produced by this current is in phase with the current. The current which voltage source V sends through field coils L_3 and L_4 must pass through high-capacity condenser C and the capacitive reactance of this condenser is so high with respect to the inductive reactance of these coils that current i_1 leads the A.C. voltage V by almost 90°. The curves in Fig. 23B give the phase relationship of the two currents, i and i_1 , and show that i_1 leads *i* by almost 90°; since a flux is always in phase with the current which produces it, these curves can also represent the flux produced by each pair of poles.

You can readily see that sine wave alternating current i, flowing through the horizontal pair of poles, will cause the flux to reach a maximum in one direction (Fig. 23C), drop almost to zero (Fig. 23D), reach a maximum in the opposite direction (Fig. 23E), and again drop almost to zero (Fig. 23F). The vertical pair of poles is going through this same complete cycle of changes in flux, but the flux is here 90° ahead of that for the horizontal poles. At each instant, the fluxes produced by the two pairs of poles combine, and it is the resultant flux which acts upon the shorted loops of wire on the armature. Referring to Figs. 23C, D, E, and F, you can see that the weaker flux in each case (shown by a single arrow line), will have little effect upon the stronger flux. Furthermore, the position of the stronger flux at each quarter-cycle is such that we have the same effect as was produced by the rotating magnetic field in Fig. 22,



FIG. 25. Typical rotor of an induction motor, made up of copper rods which are riveted and soldered at each end to circular copper diacs mounted on the rotor shaft. Before assembly, laminated sheet steel discs having slots for the copper rods are placed on the shaft inside this 'squirrel-cage'' this steel core serves to reduce the reluctance of the path for the rotating magnetic flux and for the fluxes set up by the many shorted loops in the rotor.

with our magnetic field rotating in a clockwise direction in this particular case.

A counter-clockwise rotation of the magnetic field could be secured in

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Fig. 23 simply by inserting the condenser in series with the other pair of coils, such as at point x. Since in tuning motors it is usually necessary to have a reversing switch, this particular type of induction motor will usually be found connected as shown in Fig. 24, with a single-pole, doublethrow switch arranged to change the condenser from one circuit to the other. The condenser could just as well be replaced by a high-inductance coil, making current i_1 in Fig. 23A lag rather than lead current *i*, for it is the difference in phase which creates the rotating magnetic field. The rotors used in this and all other types of induction motors are much the same in construction and appearance, resembling that shown in Fig. 25.

Shaded-Pole Induction Motor. Another method of producing a revolving magnetic field is shown in Fig. 26A. Alternating current is fed through coil L_1 , which is wound on one of the poles of the motor, with the result that an alternating flux θ_1 is produced at the pole face. A short-circuited coil L_2 is wound around one portion of the pole; when flux θ_1 is increasing, that portion of



FIG. 26. Diagrams showing principles of a shadedpole type induction motor.

 θ_1 which passes through coil L_2 will likewise be increasing. The result is that a voltage will be induced in L_2 , and this will set up a flux θ_2 which tends to oppose any change in the flux through this coil. In other words, when θ_1 is increasing, θ_2 will oppose the main flux θ_1 and the flux near point y on the pole face will be quite weak. When θ_1 is decreasing, θ_2 will attempt to prevent this decrease by aiding the main flux, and under this condition there will be a stronger flux at point y than at point x. During one-half of a cycle the region of .



Typical examples of motors used in electro-mechanical automatic tuning systems.

Courtesy Alliance Mfg. Co. Courtesy Alliance Mfg. Co. The shading coils are connected according to the diagram in Fig. 26B, and motor is therefore reversible. Note the speed-reducing gear assembly mounted directly on the motor.

Courtesy Delco Appliance Div., General Motors Sales Corp. B-Delco split-phase (reversible) induction motor; this unit employs the auxiliary high-inductance winding shown in Fig. 27 to produce the out-of-phase flux required for a rotating magnetic field. These motors are made for 5, 20 or 115 volt A.C. operation.

Courtesy Barber-Colman Co. C---Reversible single-phase, shaded-pole Barcol induction motor designed for 110-volt A.C. operation. A single main coil serves for both poles; separate shading coils are wound around a portion of each pole and connected as in Fig. 26B to permit reversal of the motor.

Courtesy Delco Appliance Division D-Delco reversible series motor. Two brushes resting on the commutator feed current to the armature windings in the correct direction. Field coil connections are as shown in Fig. 21D, permitting reversal of the motor with a simple single-pole. ...double-throw switch.

maximum flux travels from x to y; during the next half cycle this process repeats itself, with the point of maximum flux again moving from point x to point y. We thus have a maximum flux rotating always in the same direction across this pole face, touch each other, causing a shorted mica condenser. A simple ohmmeter test will reveal this trouble once the defective condenser is located.

Trimmer Condenser. As you can see in Fig. 18, there is only one trimmer condenser mounted underneath the chassis. This unit is constructed very much like the trimmer condensers already studied.

Just as with the other trimmer condensers in this receiver, the chief problem is *knowing how* to make the simple screwdriver adjustment which is required during alignment of the receiver.

This particular trimmer condenser is technically known as the oscillator padder, because it is connected in series with the tuned circuit of the oscillator. You will learn later that this trimmer is adjusted to make the tuning dial readings more accurate at the low-frequency end of the tuning dial.





FIG. 20. How a mica condenser is made.

R. F. Coils

There are two r.f. coils (radio frequency coils) under the chassis. Each r.f. coil has the basic construction shown in Fig. 21, in which there are two windings placed side by side on a waxed cardboard or bakelite coil form. These windings may either be in a single layer as shown in Fig. 21, in many layers wound one over the other in a criss-cross manner as they are on one of the coils in this receiver, or even with one coil wound right over the other.

One of the r.f. coils is known as the antenna coil, because it has a direct



FIG. 21. Construction of an r.f. coil.

connection to the antenna terminal of the receiver. Its job is to transfer the incoming modulated r.f. carrier signal from the antenna circuit to the grid circuit of the mixer-first detector tube.

The other r.f. coil is known as the oscillator coil, because it is a part of the tuned circuit of the r.f. oscillator.

These r.f. coils work hand in hand with the two sections of the gang tuning condenser to let only the one desired carrier signal get through the receiver.

R.F. coils have much the same troubles as i.f. transformer coils. Thus, they may have broken wires or poor connections at terminals. Sometimes adjacent turns of wire may touch each other and cause shorted turns. When a radio man finds that an r.f. coil is defective, he invariably replaces the coil with a new one, befrom the main pole toward the shorted coil, which is known as a shading coil. The rotor will follow this rotating flux, and consequently the rotor of an induction motor will always rotate toward a shading coil.

The direction of rotation can be reversed by placing a shading coil on each pole of the motor, with the various coils connected together in the manner shown in Fig. 26B. When switch SW is at contact 1, only the right-hand shading coil will be shorted and effective, and the rotor will rotate in a clockwise direction, toward this shading coil. When switch SW is at contact 2, the left-hand shading coil will be shorted and the direction of rotation will be counter-clockwise.

Instead of short-circuiting the shading coils, as was done in Fig. 26B, they may be connected together and to the main voltage supply. Since their reactances will be different from the reactances of the main field coils, the shading coils will produce fluxes which are out of phase with the main fluxes, thus giving a rotating magnetic field.

Split-Phase Induction Motor with Reactive Winding. Another way of producing a rotating magnetic field is that shown in Fig. 27, which involves using an extra pole on which is wound a coil (L_3) having a high inductive reactance. The A.C. line current flowing through the regular field coils L_1 and L_2 produces a pulsating A.C. magnetic flux, which alone is not sufficient to cause rotation. When switch SW is set at contact 2, placing L_3 across L_2 , the current through highinductance coil L_3 will lag the current through low-inductance coil L_2 by almost 90°, and consequently the flux produced by L_3 will lag the flux from L_2 by almost 90°. This phase relationship will be maintained as each flux increases and decreases in a sine wave manner, with the result that we have the required rotating magnetic field. When switch SW is set at contact 1, shunting L_3 across L_1 , we secure this same phase relationship between the fluxes but now the resultant or combined flux will rotate in the opposite





direction, causing reversed rotation of the rotor.

Speed. The maximum speed at which an induction motor can operate is of course the rotational speed of the rotating magnetic field. The motor can approach this speed only under no load conditions, and will slow down (due to increased slippage) as a load is applied. The maximum speed can easily be computed; the speed in revolutions per second is equal to the frequency of the alternating current in cycles per second divided by the number of pairs of poles.

The tuning motors used in radio receivers ordinarily have two poles, forming a single pair, since this construction has been found efficient as well as economical in cost. For 60cycle current, then, the speed will be 60 divided by 1, or 60 revolutions per second. To obtain the speed in revolutions per minute (r.p.m.) we multiply by 60, getting 3600 r.p.m. as the maximum speed of an induction tuning motor. If a gang tuning condenser were driven directly by such a motor, the required half-revolution would take only a little more than 1/200 of a second; obviously this is far too fast. Speed-reducing gears must therefore be used with induction motors; ordinarily they are designed to give a reduction in speed of at least 100 to 1, which means that the longest time required for a motor to tune in a station will be only a few seconds.

TEST QUESTIONS

Be sure to number your Answer Sheet 35FR-1.

Place your Student Number on every Answer Sheet.

Never hold up one set of lesson answers until you have another ready to send in. Send each lesson in by itself before you start on the next lesson. In this way we will be able to work together much more closely, you'll get more out of your Course, and you will receive the best possible lesson service.

- 1. Into what three groups (each having a different operating principle) can automatic tuning systems be divided?
- 2. What is the chief cause of oscillator frequency drift in an ordinary super-
 - 3. Into what two general groups can mechanical automatic tuning systems be divided according to the manner in which they are operated by the listener?

will a slight change in trimmer condenser capacity have more deeffect—in a preselector tuned circuit or in the oscillator tuned

the two general types of electro-mechanical automatic tuning s. Non - Homing system T Sell Homing Systems.

lectro-mechanical automatic tuning system, why is the A.F.C. eleased temporarily while the tuning motor is in motion or just stops?

audio silencing?

wo types of small A.C. motors are in general use in automatic ystems?

the direction of rotation be reversed in a series motor? direction will the rotor of a shaded-pole induction motor rotate?

29

UNDERSTANDING

"Happy is the man that findeth wisdom, and the man that getteth understanding. For the gaining of it is better than the gaining of silver.

And the profit thereof than fine gold. Understanding is more precious than rubies: And none of the things thou canst desire are to be compared unto it.

Its ways are ways of pleasantness, And all its paths are peace. It is a tree of life to them that lay hold upon it.

And happy is every one that retaineth it."

-Adapted from Proverbs.

The world is yours. Stored in your subconscious mind is knowledge acquired through your own efforts, plus abilities bequeathed by all the centuries of the past. Once you truly understand the tremendous capacities which are within you—once you realize your responsibility to others in the world once you acquire the knowledge needed for success in your own chosen field—you can be as great as anyone who has ever lived before you!

J. E. SMITH

cause it is difficult to make satisfactory repairs with the very fine wire used in winding these coils.

Power Transformer

When you see a large, heavy unit like that in Fig. 22 on top of the chassis of a radio receiver, you can be sure it is the *power transformer*.

All connecting wires for the power transformer in our receiver are underneath the chassis. The wires have various colors for purposes of identification.



FIG. 22. The power transformer in a radio set is a heavy, bulky unit. It is covered with a metal cap for two reasons, to protect prying fingers from the high voltages which exist in the coils of wire inside, and to prevent the magnetic effects inside from escaping and interfering with other parts.

Construction. A power transformer consists of a number of separate coils of insulated wire, wound one over the other on a core made from thin strips of soft steel stacked together.

Purpose. The power transformer in a radio set transforms the a.c. power line voltage to the various higher and lower a.c. voltage values which are required by the tubes in the receiver. The more tubes there are, the larger must be the power transformer if it is to supply these tubes with the required electrical power without getting too hot.

Possible Troubles. There are a number of defects in receiver circuits which can place an *excessive load* on the power transformer Overloading causes a power transformer to get hot, and this excessive heat damages the insulation between the windings of the power transformer.

When the insulation on the wires or between the coils of a power transformer loses its insulating qualities because of heat, adjacent turns touch each other and cause a short circuit. This allows more current to flow, and produces still more heat. The condition becomes worse and worse. Eventually the trouble becomes so bad that the wire in the transformer melts and opens up at some point, causing complete failure of the receiver.

A radio serviceman often relies upon his nose to tell him when a power transformer is bad. Failure of this part is usually due to the overheating just described. The resulting smell of burned insulation is easily recognized, hence a radio man suspects the power transformer first of all whenever he encounters this strong, unpleasant odor of burned insulation.

An ordinary break in one of the wires inside a power transformer is rather rare, and yet may happen occasionally. Either voltmeter or ohmmeter measurements at the power transformer would reveal this trouble. Poor connections at the power transformer terminals would have a similar effect.

Defective power transformers are replaced rather than repaired. The replacement offers no difficulty if



FIG. 23. Additional under-chassis parts of our superhet receiver are identified in this view.

you make a careful diagram of the original connections before disconnecting the damaged transformer.

Other Under-Chassis Parts

As you can see in Fig. 23, there are still quite a few parts underneath the chassis of our receiver which have not yet been taken up. Most of these are highly important parts which often require attention by a radio serviceman, so a brief study of each will be well worth while.

Power Cord. One trouble which occasionally baffles even the most expert radio man because of its very simplicity and obviousness is a *defective power cord*. This cord shows clearly in Fig. 25, and serves to connect the receiver to an a.c. power line. The cord has a standard wall outlet plug at one end, and the two leads at the other end connect to two points underneath the chassis of the receiver.

The most likely places for trouble to occur in a power cord are at the plug, at the receiver connection, and at the point where the cord passes through a hole in the chassis. Examine all these places carefully for poor connections or damaged insulation.

The rubber used in a power line cord ordinarily lasts only about five years. Therefore, whenever you encounter an old receiver having brittle or cracked rubber insulation on the power line cord, it is best to install a new cord. The replacement job is quite simple, as it involves only the resoldering of two connections.

Always examine the power cord plug carefully whenever you work on a receiver, to make sure that there are no loose strands of wire which might eventually touch the opposite terminal of the plug and cause a short which would blow the fuses in the house. Tighten the screws in the plug if they are loose.

On-Off Switch and Volume Control. As you already know, a single control knob serves for both of these parts. The on-off switch is mounted at the back of the volume control. The switch is connected in series with one of the power cord wires, so that it can open the power supply circuit. The on-off switch rarely gives trouble. About the only thing you need watch for is a poor connection at either of the two switch terminals.

The volume control is definitely a trouble-maker, however, in any radio receiver. In this set, it is in the circuit of the second detector, and governs the strength of the audio signal which is fed into the first a.f. stage by the second detector.

The volume control in this receiver is a type of variable resistor which consists of a strip of carbon resistance material, and a rotating contact arm which sweeps over the carbon material. The contact arm can be set to make contact at any desired point along the resistance material.

If the carbon material in a volume control wears away due to repeated use, or if the contact arm becomes loose, we have the equivalent of a poor connection. The radio man says the volume control has become noisy, because a noise is heard from the loudspeaker every time the control is adjusted.

The best remedy for a noisy volume control is replacement with a new volume control. It is usually customary to order a new switch at the same time, since the two are furnished as a single unit.

If you make a rough picture diagram of the connections to a defective volume control before removing it, no trouble should be experienced, in making connections to the new unit. Even experienced servicemen follow this procedure, to reduce chances for mistakes.

Shielded Wire. One of the leads going to the volume control in our receiver is enclosed in a tube of braided wire which serves as a shield. This shield is necessary to prevent that

particular wire from being affected by stray electric and magnetic fields around the chassis. Without this shielded wire, the set might have an annoving squeal or hum.

Tone Control. The tone control circuit in this receiver consists of a variable resistor connected in series with a condenser in the audio output stage. The condenser side-tracks the higher audio frequencies, and the variable resistor determines how much the higher frequencies will be side-tracked or suppressed. This variable resistor thus controls the tone of the program coming from the loud-speaker.

The tone control resistor is less likely to cause trouble than the volume control, but the two parts are similar and do have the same types of trouble.

Dial Lamp. Mounted behind a hole in the front of the chassis is a small socket containing a *dial lamp*, also known as a *pilot lamp*. This lamp illuminates the tuning dial when the receiver is turned on, and thus also serves to indicate whether or not the receiver is turned on. The dial lamp in our receiver requires a voltage of 6.3 volts, the same as the filaments of the tubes in the set. The lamp is therefore connected in parallel with the tube filaments.

There is only one wire connection to the dial lamp and to each tube filament, because the circuit to the other terminal is in each case completed through the metal chassis. This arrangement is common practice in radio sets, and is illustrated in Fig. 24.

A dial lamp has about the same life as an ordinary electric light bulb in a home, and hence requires replacement occasionally. The easiest way to insure getting the correct new lamp is to take the old bulb to the radio supply store and ask for a similar new bulb.

When ordering a dial lamp by mail from a catalog, select one having the same type of base, the same voltage rating, and the same color of glass bead inside the lamp as that in the original lamp. The color of this glass bead is a clue to the current required by the lamp.

In some receivers, the color of the glass bead is unimportant, but in others the wrong pilot lamp can defious terminals of the old socket before you remove it.

When sockets are riveted to the chassis, they can be removed by drilling out the rivets. The new socket can be mounted with bolts and nuts or with new rivets.

Conclusion. Remember that failure of a radio receiver is due to some simple breakdown in one or more of its parts. The breakdown is purely mechanical, such as a broken wire, defective terminal or parts touching each other. Usually the breakdown



FIG. 24. Filament circuit, with dotted lines indicating the connections which are completed through the chassis of the receiver. All tube filaments and the pilot lamp require 6.3 volts, and the transformer secondary provides this voltage.

nitely cause tubes in the receiver to burn out, as you will learn later.

Tube Sockets. The five tube sockets on the chassis of this receiver are of modern design, and rarely if ever give trouble. However, on older sets you will occasionally encounter tube socket contacts which do not grip the tube prongs tightly enough. If it is impractical to tighten the socket clips sufficiently, a new socket must be installed. The installation of a new tube socket should offer no difficulty to you even right now, if you make a careful sketch showing how connections were made to the varican be seen if you go to the trouble of taking the defective unit completely apart.

Of course, it is impracticable to take apart every unit in a receiver when hunting for trouble. You won't have to do this, because the knowledge of how radio parts work in circuits and the radio servicing techniques you learn in the N.R.I. Course will enable you to locate defective parts speedily with simple test instruments. In some cases, you will even be able to spot the guilty part by noting the manner in which an ailing receiver is operating.

World Radio History

Radio Receiver Servicing Techniques

If you have twelve identical radio receivers all with the same trouble, and give these receivers to twelve different radio servicemen for repair, the chances are pretty good that each man will use a different method for locating the trouble!

This is an entirely normal condition, because radio servicing techniques depend a great deal upon the amount of training and experience a man has. All twelve men will undoubtedly find the trouble, but some will find it much faster than others, because their methods are more efficient.

To illustrate some of the methods which could be used to find trouble in a receiver, let us imagine that a defective receiver has come into your radio shop for repair.

Check of Performance. All servicing methods start with a thorough check of performance, to see just how the radio receiver is misbehaving. Oftentimes this check gives valuable clues to the source of trouble.

To make a performance check at a service bench, you plug the receiver power cord into a wall outlet, make antenna and ground connections if the set does not have a built-in aerial, then turn on the receiver and attempt to tune in stations. You note how the volume control, the tone control and any other controls affect performance. Your training and experience have taught you to pay particular attention to unusual squeals, noises or hum, because each tells its own story of trouble.

Testing Tubes. Since bad tubes are common causes of trouble in receivers and since tubes are easily tested, your next step is a test of each tube in the receiver. Tubes should be tested at some point in the servicing procedure anyway, so this might as well be done first of all since it is so easy to do.

A beginner might prefer to check the tubes later if he didn't have a tube tester, for he would usually have to take the tubes to some reliable store for testing.

An expert serviceman might be able to figure out where the trouble is just by listening to the receiver, and clear up the trouble before making a routine check of the tubes. In the great majority of cases, however, you will find that tube testing comes first on the serviceman's list of trouble-hunting techniques.

Your aim in servicing is to narrow down the trouble first to a section of the receiver, then to a stage, and finally to just one circuit and to a particular part in that circuit.

The average superheterodyne receiver can be divided into six sections for servicing purposes. The preselector section includes all circuits between the antenna and the mixer-first detector. The frequency converter section includes the local r.f. oscillator stage and the mixer-first detector stage. The *i.f.* amplifier includes all i.f. stages and i.f. transformers. The second detector is just one stage. The audio section covers everything following the second detector, including the loudspeaker. The power pack section includes the rectifier tube and its associated parts.

Sometimes an experienced, properly trained serviceman can tell the section containing the trouble just by listening to how the receiver is misbehaving. If the faulty section has more than one vacuum tube stage, he then makes simple tests which isolate the defective stage.

Other servicemen prefer to start right in with tests which locate the defective stage. Let us select just two of these tests now, and see how you would carry them out.

Circuit Disturbance Test. For a dead receiver, the simple circuit disturbance test is one of the fastest and most effective of these tests for locating the defective stage. It is carried out by introducing an electrical disturbance in each tube circuit in turn while the receiver is turned on.

You start at the audio output stage (next to the loudspeaker) and work backward through the stages when carrying out the circuit disturbance test. The disturbance which you introduce causes a click or thump in the loudspeaker when all the stages between the point of disturbance and the loudspeaker are good, but no click is heard when you arrive at the defective (dead) stage.

The electrical disturbance for this test can be inserted in a stage by removing a tube momentarily, by touching the top cap of the tube if there is one, by removing and replacing the top cap, or by shorting the grid of the tube momentarily to the cathode of the tube.

Since you must go from tube to tube in a definite order if a circuit disturbance test is to be effective, one requirement for making this test is the ability to identify the various tubes on top of the chassis. An entire lesson later in your Fundamental Course is devoted to this highly practical problem of finding your way around in a receiver chassis.

Stage-Isolating Procedure. For some receiver troubles, such as low volume, you introduce a test signal in one stage after another, working toward the antenna. Again the loudspeaker tells when you have come to the defective section or stage, because the introduced signal will increase in volume for good stages but not when you move through the defective stage. For each other type of receiver complaint, there are one or more professional techniques for tracing the trouble.

Locating the Defective Part. Having located the defective stage in our ailing receiver, you could now follow the practice of some servicemen and begin testing the parts in the defective stage, one after the other. This eventually locates the defective part, but it can take a lot of valuable time.

As a professional radio serviceman, however, you would make additional general tests which gradually narrow the trouble down to a particular circuit in the stage. Then, by testing only a few radio parts at the most, you would locate the defective part in the quickest possible time.

It is in locating the defective stage and circuit that we find such a great variety of servicing techniques. There is no one technique which can be considered the best for all jobs; each has its own advantages, and will work best in a particular job. If you choose radio servicing for your advanced study, you will learn these servicing techniques and will also learn when to use them most effectively.

Short Cuts. After years of experience in radio servicing, a man learns to associate certain ailments and symptoms of a receiver with definite defects. He can then take short cuts through the general servicing procedure which bring him more rapidly to the defective part. One important requirement for taking any short cuts, however, is a thorough knowledge of radio fundamentals. Unless these are thoroughly mastered, radio servicing becomes an unprofitable guess-andtry proposition.

Guess-Work Won't Pay. If you consider how many different tubes, resistors, coils, condensers, transformers, connections and other parts there are in a radio receiver, the folly of expecting to service radio receivers simply by testing one part after another without thinking becomes clearly evident.

You Can Start Soon. Once you learn the fundamental facts about radio parts and circuits as presented in the early lessons, and learn how to perform some of the simpler tests and servicing techniques as outlined in your practical N.R.I. Job Sheets. you will be in a position to start the N.R.I. practical training plan for securing radio receiver servicing experience right in your own home. After completing this training plan according to our instructions and mastering the essential radio information given in Fundamental Lessons, you will be able to do service work nearly as well as the average radio technician.

By the time you have finished your N.R.I. Course and followed all our experience-getting suggestions, you will be way ahead of the average radio technician. You will be using speedy, modern servicing techniques which allow you to fix any radio set whatsoever, and which enable you to handle

much more work than poorly-trained competitors can do in a similar period of time. Yes, N.R.I. graduates definitely get more business and make more money than less well-trained men.

After a defective part is located and replaced. the wise serviceman checks performance to make sure that there are no other defective parts. In addition, he readjusts or rather aligns the receiver for best performance, and cleans both the chassis and cabinet so as to restore the receiver as nearly as possible to its new condition. This final check-up and clean-up can prevent future trouble, and is definitely appreciated by customers. These extra services help to justify the fair charge which servicemen should make for their knowledge and experience.

Looking Ahead

Having now completed the groundwork for your study of radio, you are ready to add to your knowledge of how radio parts work and what they do.

In the next lesson you will study different types of resistors, and learn how they are used in practical radio circuits to control voltage and current. The two following lessons will cover coils and condensers in the same thorough manner.

This next group of three lessons will thus complete your knowledge of three important parts used in radio circuits.

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Lesson Questions

Be sure to number your Answer Sheet 4FR-2.

Place your Student Number on every Answer Sheet.

Send in your answers for this lesson immediately after you finish them, as instructed in the Study Schedule. Do this, and get the greatest possible benefit from our speedy personal grading service.

- 1. If there is no set screw on the control knob of a receiver, how would you remove the knob?
- 2. Why is a shield placed around one of the tubes in our superheterodyne receiver?
- 3. Name three ways in which radio tubes commonly fail.
- 4. If a radio receiver sounds distorted, and you can hear a rubbing sound when you push the loudspeaker cone in and out with your fingers, what is the trouble?
- 5. You are tuning a radio receiver towards one end of the dial. The set suddenly goes dead as you tune, and all you can hear is the noise caused by the gang tuning condenser plates scraping against each other. How would you correct this trouble?
- 6. State briefly the purpose of the small trimmer condensers which are mounted on top of the gang tuning condenser.
- 7. What type of meter is ordinarily used to locate shorted or leaky paper condensers?
- 8. What happens to a power transformer when you overload it by drawing too much current from it?
- 9. How many connections must be unsoldered in order to disconnect the power cord of an a.c. superheterodyne receiver like that studied in this lesson?
- 10. At what stage do you start, when using the simple circuit-disturbance test in a dead receiver?

World Radio History

THE VALUE OF COURTESY

A recent survey showed that people complain more about discourteous clerks than about any other fault a business could have. In fact, many people pay extra at higher-priced stores just to get the courtesy and respect they feel entitled to.

Regardless of whether you work for someone else or have a radio business of your own, plain ordinary courtesy can bring many extra dollars to you.

Genuine courtesy comes from the heart. When you say "thank you," be sure you sincerely mean it.

Courtesy becomes a habit if practiced long enough. Be courteous to everyone—to members of your family, to those who don't buy from you, even to the very lowest persons who serve you—then you can be sure you'll be courteous when it really counts.

Give your courtesy with a smile. There is an old Chinese proverb which says, "A man who doesn't smile shouldn't keep a shop." And you're keeping a "shop" even if you are selling only your ability and knowledge to an employer. A friendly smile is itself courtesy of the highest type, bringing you unexpected returns in actual money as well as friendship.

World Radio History

J. E. SMITH



STUDY SCHEDULE NO. 5

Use this schedule to help you master the fifth part of your N.R.I. Course. Unless other instructions are given, study one step at a time, as follows: Read the specified pages first at your usual reading speed, then reread slowly one or more times until you understand the entire section. Finish with one quick reading to tie the ideas together in your mind, then figure out your answers to the Lesson Questions specified for that step.

2. Quickly review pages 3 to 13 of Lesson 3FR-2. These pages contain three sections: What Radio Men Measure; How Voltage

- These pages contain three sections: what hadio Men Measure; How voltage and Resistance Affect Current in Simple Circuits; Load Connections for Radio Circuits. It is highly essential for you to review this material, so do it now. One quick reading should be enough.
- 3. Ohm's Law Tells "How Much" for Resistors. Pages 2 to 6 of this lesson Ohm's Law, the radio man's most important rule for figuring out what has happened, is happening or will happen in a circuit containing resistance, is presented here in a complete, easy-to-understand manner. After you master this section, answer Lesson Questions 1, 2, 3, and 4.

- □ 7. How Resistors Are Made......Pages 21 to 28 The characteristics, advantages and disadvantages of various types of resistor construction are taken up. Answer Lesson Question 10, then read "Looking Ahead" on page 28.
- 3. Mail Your Answers for Lesson 5FR-4 to N.R.I. for Grading.
- □ 9. How To Restring Dial Cords and Set Push Buttons....RSM Booklet No. 5 This is another of the series of Booklets that are to give you practical, how-to-do-it training in fixing radios. Read this Booklet now, then keep it handy for future reference. The "shop training" you get from these RSM Booklets is going to prove valuable to you throughout your service career, so you will want to read and review these Booklets many times. NOTE: Some of the Study Schedules in future Lessons may refer to "Job Sheets" instead of RSM Booklets. Read the correspondingly numbered RSM Booklet in such cases. These Booklets are an improved and enlarged version of the older Job Sheets, and are replacing them.
- □ 10. Start Studying the Next Lesson on Coils.

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World Radio History

RADIO RESISTORS AND HOW THEY ARE USED

Why Resistors Are Important in Radio

IN addition to tubes, there are only three kinds of radio devices which control voltages and currents in radio circuits, and thereby make radio possible. These are resistors, coils and condensers. Terminals, wires, moving contacts or other mechanical parts merely provide connections or supports for these three basic radio parts. You'll find resistors in power packs, reducing the high d.c. output voltage to the various lower voltage values required by the tubes in transmitters or receivers.

You'll often find resistors in tone control circuits, working with condensers to give you a choice of tonemellow, brilliant or deep bass.



Thus, volume controls are combinations of *resistors* and mechanical parts. Audio, power and r.f. transformers are combinations of *coils*. I.F. transformers are combinations of *coils* and *condensers*. Loudspeakers are combinations of *coils* and mechanical parts.

Jobs for Resistors. You'll find resistors in practically every radio tube circuit. Sometimes they work alone or work with coils and condensers to keep radio signals in their correct paths. Sometimes they furnish the correct d.c. operating voltages for the tube electrodes. Sometimes they serve as loads for radio tube circuits. In dozens and dozens of other places in radio equipment, you'll find resistors working silently and effectively to help make possible the magic of modern radio. Certainly, resistors deserve every bit of the attention we shall give them.

A Trio of Lessons. This lesson gives you practical information about resistors and the laws which govern their performance in radio circuits. The next two lessons give you the important basic facts about coils and condensers. Thus, in just three lessons you master the fundamental facts about three important radio parts.
Ohm's Law Tells "How Much"

Radio's Simplest Resistor Circuit. In Fig. 1 is the simplest complete circuit we can have in radio apparatus. The source provides exactly the correct voltage value for the resistance load, so the source and load are connected together directly by wires.

This simple circuit has only three values—the source voltage, the current, and the load resistance. In a previous lesson, you learned that these values are related to each other in a logical way. For instance, the current can be increased either by raising the source voltage or lowering the load resistance. Likewise, the current can be reduced by lowering the voltage or raising the resistance.

In most radio jobs, it is sufficient to know only in this general way how the current is going to change when either the voltage or resistance are changed. There will be times, however, when you want to know *exactly* how much the new current value will

be.



The basic radio law which gives you this definite information is known to all radio men as Ohm's Law, because it was first discovered and proved to be true by the scientist

George Simon Ohm, in 1826.

Ohm's Law expresses the exact relationship between voltage, current and resistance for any circuit containing only resistors. You'll find later that you can also make this law work for other radio parts, too.

Ohm's Law for Current. To find the current, divide the voltage value by the resistance value.

The voltage value should be in volts

(as it usually is), and the resistance value should be *ohms*. The result will then be the current in *amperes*. For example, if the voltage is 50 volts and the resistance is 25 ohms, the current will be 2 amperes $(50 \div 25 = 2)$.

Formula for Ohm's Law. Radio men usually prefer to work with a simplified version of Ohm's Law, in



FIG. 1. Simplest basic circuit used in radio. The source voltage is exactly correct for the load, so these two parts are here connected together directly by wires. When the source voltage is too high for the load, however, the connection is made with a combination of wires and radio parts, chiefly resistors. The connecting system is sometimes known as a TRANS-MISSION SYSTEM, because it serves to "transmit" power from the source to the load.

which letters and arithmetic signs replace words. These letters are an important part of the language of radio, so let us get acquainted with them now.

R is always used to represent RE-SISTANCE.

I is always used to represent CUR-RENT. (This may not seem logical, but is done to avoid confusion because the letter C is used for condensers. If you like, you can think of I as standing for *intensity of current*.) E or V is used to represent VOLT-AGE. (E comes from *electromotive force*, which is another name for a source voltage.)

By using these letters, we can condense Ohm's Law into a simple formula which is much easier to use.

Instead of saying that current (I) is equal to voltage (E) divided by resistance (R), we can simply say: I = E \div R.

You will occasionally find Ohm's Law expressed in an even simpler manner. One letter is placed under the other to indicate division. The short line between the letters means that the value above the line is to be divided by the value below the line. This dividing line can either be slanting or horizontal. When we express Ohm's Law in this simplest possible form, here is what we get:

$$I = \frac{E}{R}$$
 $I = E/R$

Both formulas are pronounced I equals E over R, and both mean the same thing—that the amount of current is equal to the amount of voltage divided by the amount of resistance.

Ohm's Law can be used to find the exact current value flowing through a complete d.c. circuit or through any portion of the circuit. Of course, if you are figuring the current in a complete circuit, all your values should be total values for the entire circuit. Likewise, when figuring the current in part of a circuit or in a single resistor, all your values must be for just that part of the circuit. A few simple examples will illustrate these entirely logical rules for applying Ohm's Law.



Suppose we want to find the current in this series circuit. The voltage which acts on the *complete* circuit is

the battery voltage: let's say it is 90



Here are just a few of the resistor questions which are answered in this lesson.

volts. The total circuit resistance is the sum of the individual resistor values since all four resistors are in series; let's say these values are known and add up to 30 ohms. The current value is equal to the voltage value divided by the resistance value, so we divide 90 by 30. This gives 3 amperes as the circuit current.

Now suppose we need to find the current through the single cathode resistor in this tube circuit, and haven't time to unsolder the cathode connection and insert an ammeter. We measure the voltage across



the resistor with a d.c. voltmeter let's say we find it's 12 volts. Next, we read the resistance value marked on the resistor—and let's say this is 300 ohms. Remembering $I = E \div R$, we divide 12 by 300, and get .04 ampere as the current through the resistor. Notice that we didn't have to pay any attention to the rest of the



No matter whether it's the voltage, current or resistance value you want to know for any one of the resistors underneath the chassis of this receiver, Ohm's Law can tell you how much the value will be

circuit; we wanted to know the resistor current, so we used only the voltage and resistance values of the resistor itself.

One thing you must realize when using Ohm's Law is that you are dealing with numbers. You are not dividing volts by ohms. You can't do that any more than you can divide dollars by people. However, you can divide a *number* of dollars by a *num*ber of people, to get something entirely different-the number of dollars each person gets. That's what we do in radio-we divide the number of volts by the number of ohms, and get something entirely different-the number of amperes flowing. Keep this in mind, and you will never be asking, "How many volts are there in an ohm?" This question would be just like asking how many people there are in a dollar!

Finding Voltage or Resistance. So far, we have used Ohm's Law only to find the current when we know the voltage and resistance values. But sometimes we know what the current is, and want to find one of the other values instead. This can still be done with Ohm's Law—in fact, whenever any two of the three values (voltage. current and resistance) are known, Ohm's Law will help you find the third value.

It is perfectly possible to use our already-familiar current formula to find voltage or resistance, even though this might be a bit inconvenient. For example, if I was 2 amperes and E was 10 volts, but R was unknown, we could start with I = E/R. Putting our two known values in place of the letters, we would get 2 = 10/R. This tells us that 10 divided by the value of R gives 2. There is only one value of R which will work here—5. Thus, we find that the resistance is 5 ohms.

In much the same way, the current formula could be used to find the value of E. It works out all right, but in many cases it is easier and quicker to use Ohm's Law in two other forms, one for finding voltage and the other for finding resistance. While you are reading about these, keep in mind that the three versions of Ohm's Law are just convenient different ways of expressing the same basic principle.

Ohm's Law for RESISTANCE. To find the resistance, divide the voltage value by the current value. Thus, if the voltage is 10 volts and the current is 2 amperes, the resistance will be 5 ohms $(10 \div 2 = 5)$.

This version of Ohm's Law can likewise be expressed in simpler forms, as $R = E \div I$ or R = E/I.



The filament circuit in a portable radio receiver makes an interesting example for this resistance-finding f or m u l a. The voltage is 1.5 volts, and let's say the rated filament current is .1 ampere for the tube shown here. First we write: $R = E \div I$. Next, we substitute the two known values: R =

 $1.5 \div .1$, and divide, thus getting 15 ohms as the resistance of the filament when hot.

Ohm's Law for VOLTAGE. To find the voltage, multiply the current value by the resistance value. Thus, if the current is 2 amperes and the resistance is 5 ohms, the voltage will be 10 volts $(2 \times 5 = 10)$.

This version of Ohm's Law can be written as $E = I \times R$. Sometimes the multiplication sign is omitted, giving E = IR. (In formulas, letters are often placed side by side like this to indicate multiplication.)

To illustrate a practical example of this voltage-finding formula, we can take the plate circuit of almost any vacuum tube and draw this one complete circuit by itself as shown in Fig. 2. We want to find out how much d.c. voltage to expect across plate load resistor R. All we do know is that this is a 50,000-ohm resistor with a



FIG. 2. Drawing the plate circuit of a tube by itself in this way simplifies any Ohm's Law figuring you might want to do for the resistors in the circuit.

current of .002 ampere flowing through it. These two values are all we need to use the formula $E = I \times R$. We get $E = .002 \times 50,000$. Multiplying this out gives 100 volts as the voltage drop which would normally exist across R.*

What About A.C. Circuits? Ohm's Law holds true for all simple d.c. circuits and for all a.c. circuits which contain only resistors, provided that voltage values are in *volts*, current values are in *amperes*, and resistance values are in *ohms*.

^{*} As a special feature of the NRI Consultation Service, a Data Sheet on resistors will be sent to you with your graded answers for this Leeson. This Data Sheet contains a number of additional examples of applying Ohm's Law to practical resistor circuits, as well as a reference listing of all the formulas taken up in this Lesson.

You are not required to study this supplementary material; it is provided only as additional information for those who want to learn more about this subject. Remember, you won't have to do any involved calculating when you do radio servicing—these examples are given to help you understand what happens when parts of different resistances are used.

Useful Data for Ohm's Law

In radio, current values are often in milliamperes, and resistance values are occasionally in megohms. Once in a while, you will even encounter millivolts and microvolts instead of volts. Before you can use these values with Ohm's Law, you must change them to amperes, ohms and volts.

Changing Milliamperes to Amperes. You will recall from a previous lesson that a milliampere is a thousandth of an ampere. Therefore, a value in milliamperes must be divided by 1000 to change it to amperes. This is easily done by moving the decimal point THREE places to the LEFT. Here are a few examples:

.3	milliampere =	.0003	am per e
1	milliampere =	.001	am per e
25	milliamperes =	.025	ampere
600	milliamperes =	.6	ampere
1000	milliamper es =	1.	am pere

Changing Megohms to Ohms. In radio work, we often deal with resistance values running up into millions of ohms. To avoid having to write large numbers a larger unit of resistance called the *megohm* is used. One megohm is equal to one million ohms.

To change a resistance value in megohms into ohms, all you need to do is multiply by 1,000,000. The easiest way to do this is by moving the decimal point SIX places to the RIGHT. Here are some examples:

.1	megohm	=	100,000	ohms
.75	megohm	=	750,000	ohms
1	megohm	=	1,000,000	ohms
2.5	megohms	=	2,500,000	ohms
10	megohms	=	10,000,000	ohms

Changing to Volts. A millivolt is one thousandth of a volt, so a value in millivolts must be divided by 1000 to change it to volts. This can be done by moving the decimal point THREE places to the LEFT.

A microvolt is a millionth of a volt.

A value in microvolts must be divided by 1,000,000 to change it to volts. This can be done by moving the decimal point SIX places to the LEFT.

Symbols for Ohms. Resistance values appear so many times in radio diagrams that a single Greek letter is, for convenience, used instead of the word *ohm*. The two ways of writing this letter, or rather symbol, are shown in Fig. 3. You will find that



FIG. 3. Symbols used in radio for ohm. (The letter is omega, pronounced oh-MEE-gah. It is the last letter of the Greek alphabet.)

some diagrams use one way, some the other. On your own diagrams, you can choose whichever symbol is easier for you to make, because both are perfectly correct.

In rare cases, you may find both symbols appearing on the same diagram. The small letter ω then represents ohms, and the capital letter Ω represents megohms. Remember this applies only when both symbols appear on the same diagram.

Abbreviations. On circuit diagrams the abbreviation MEG. is often used for megohms. Thus, a 10-megohm (ten million ohm) resistor would be marked 10 MEG. on a diagram.

You will sometimes find either the letter M or the letter K used to represent thousands of ohms. Whenever you see an M or K, just read it as thousand ohms. Thus, 250M and 250K would both be read as two hundred and fifty thousand ohms.

Figure 4 shows some of the most commonly used ways of specifying a typical resistance value on diagrams.

Using Resistors to Reduce Voltage

A Common Radio Problem. When the source voltage in a d.c. circuit is *higher* than is required by the load, the source cannot be connected directly to the load with wires, because the excessively high source voltage would send too much current through the load and possibly damage it.

"Why not change the d.c. source voltage, or get a different load which can handle this higher voltage?", you ask. Either is a perfect solution, but unfortunately they are not always possible. The high source voltage may be needed for other circuits, and the particular load may be essential for satisfactory operation of the radio set. In many cases, the only practical solution is to place between the source and load some part which will take part of the source voltage, thus lowering the load voltage to the required value.

A Resistor Will Do It. If we place a resistor between source and load in the manner shown in Fig. 5, the current flowing through the resistor will



FIG. 4. You may find resistor values specified in any of these ways on circuit diagrams used in radio work. Don't try to memorize these ways now, because you can easily figure them out when necessary, and need refer to this diagram only when in doubt. produce a voltage drop across it. This voltage drop cannot act on the load, so less voltage is available for the load. By choosing the proper resistance value for this series resistor, we can reduce the load voltage to any desired value.

A series resistor is often called a *voltage-dropping resistor*, because it uses up, wastes or "drops" the undesired portion of the source voltage.

A series voltage-dropping resistor can be anywhere in the circuit—next to the source, next to the load, on the



FIG. 5. Basic radio circuit in which the source voltage is too high for the load.

other side of the source, on the other side of the load, or anywhere in between source and load—because the same current flows through all parts of a series circuit.

There are two ways to look at this series resistor. First, we can think of it as increasing the total circuit resistance. (When resistors are in series, their ohmic values add together.) Increasing the circuit resistance reduces the current to the correct value for the load. When a load gets its correct current, it is also getting its correct voltage.

Here is the second way to think of this series resistor. Whenever current flows through a resistor, the voltage drop produced across this resistor is equal to the current value multiplied by the resistance value (E = $I \times R$). The load voltage is then equal to the source voltage minus the voltage drop across the resistor. The important thing to remember is that a series resistor of the correct value will reduce the load voltage in a simple series circuit to its correct value.

Can Voltage Vanish? In the circuit of Fig. 5, we have three different voltages: 1. The source voltage; 2. The voltage drop across the series resistor; 3. The load voltage, which is also a voltage drop.

The relationship between these voltages is highly important because it tells you how much voltage to expect in actual radio circuits. Before we take up these voltage relationships, however, study the circuit in Fig. 5 a bit and see if you can answer the following questions from reasoning and what you already know:

• Can the load voltage ever be higher than the d.c. source voltage?

• Can the sum of the two voltage drops ever be HIGHER than the source voltage?

• Can the sum of the two voltage drops ever be LOWER than the source voltage?

• Can any of the source voltage just vanish?

Kirchhoff's Voltage Law. If you answered "no" to all four questions you were entirely correct, and already have a clear understanding of the highly important voltage law now to be taken up. This fundamental law is known as *Kirchhoff's Voltage Law*, in honor of the scientist who first proved that it always holds true.

KIRCHHOFF'S VOLTAGE LAW

In a complete circuit, the sum of the voltage drops is always equal to the source voltage. **Examples.** Like Ohm's Law, Kirchhoff's Law has a great number of practical applications in radio circuits. Examples involving real radio problems always help in remembering important ideas, so let us consider now a few of the ways in which this law is used by radio men. These examples will demonstrate even more clearly that the law is "just what you would naturally expect."

Filament Circuit. In the simple filament circuit shown in Fig. 6, the storage battery sends current through the tube filament and series resistor R. If you measured each voltage in the circuit with voltmeters connected



FIG. 6. Example of a circuit in which the source has too high a voltage for the load. Series resistor R cuts the load voltage down to the required value of 5 volts.

as indicated in the diagram, you would find that adding the filament voltage drop of 5 volts to the resistor voltage drop of 1 volt gives the source voltage value of 6 volts, just as predicted by Kirchhoff's Voltage Law.

Nearly every universal a.c.-d.c. radio receiver has a series resistor in its filament circuit, to use up that portion of the power line voltage which is not needed by the filaments in series. Later in this lesson, you will get acquainted with the line cord resistors and resistance tubes which are used for this purpose in a.c.-d.c. receivers.

Plate Circuit. Kirchhoff's Voltage Law holds true for any complete circuit no matter how many other cir-



FIG. 7. This portion of an actual receiver circuit diagram is given here only to illustrate how voltages can be checked in one complete circuit regardless of nearby circuits. Note: No dots at cross-overs mean no connections.

cuits or radio parts are connected to various points in the circuit. For example, if a radio man were to pick up the service manual of an Airline Model 93BR-462A portable receiver and trace along the plate supply circuit of one tube in the manner shown by the shading in Fig. 7, he would ignore all the other circuits, and apply the voltage law just to that one circuit.

Thus, if he took a d.c. voltmeter and measured in turn the voltage across resistor R_1 , resistor R_2 and platecathode resistance R_P while the receiver was in operation, then added these three voltage drops together, the result would be equal to 90 volts, the voltage of the B battery which is the source in this circuit.

The plate-cathode path through the tube in Fig. 7 is marked R_{P} and considered as a resistance because it acts exactly like a resistor in this circuit. In other words, it obeys Ohm's Law. To find the value of this resistance,

you would divide the plate-cathode voltage by the plate current ($R = E \div I$). You'll learn more about this when you study the lesson on tubes.

Note that numbers and letters are used after the letter R to distinguish between the three resistors in this plate circuit. This is always done whenever more than one resistor is shown on a diagram. You can pronounce these notations as R sub one and R sub pee, or simply as R one and RP.

Variable Series Resistors. In some circuits, we use a variable resistor or rheostat in series with the load, and adjust its resistance value until the load is getting its correct voltage and current.

The construction of one type of variable resistor is shown in Fig. 8. The position of the contact arm can be changed by rotating the shaft of the unit. Since the movable arm is connected to one terminal, the position of the arm determines how much resistance there is between the terminals.



FIG. 8. Adjustable resistor or rheostat.

Resistors in Parallel

When two resistors are connected in parallel to a voltage source, as they are in Fig. 9A, both resistors are acted on by the same voltage. This means that each of the parallel resistors gets the same current as if it were connected all by itself to the battery (Figs. 9B and 9C).

The current through each parallel resistor will be determined by Ohm's Law: $I = E \div R$. Thus, the current I_1 through resistor R_1 in Fig. 10 will depend upon the source voltage and the ohmic value of R_1 . Likewise, the current I_2 through R_2 will depend upon the source voltage and the ohmic value of R_2 .

Since the source in Fig. 10 must

supply current to two resistors in parallel, it is logical to expect that the source current I will be the sum of currents I_1 and I_2 flowing through the parallel resistors. This is quite correct, and we express it as a fundamental radio rule known as Kirchhoff's Current Law. Here it is:

KIRCHHOFF'S CURRENT LAW

The currents flowing toward a terminal in a radio circuit must equal the currents flowing away from that terminal.



FIG. 9. These photos illustrate the basic fact that when resistors are connected in parallel to a voltage source, each draws the same current it would get if connected by itself to the source.



In the circuit of Fig. 10, this means that current I flowing toward terminal a must be equal to the sum of currents I_1 and I_2 flowing away from terminal a. Likewise, $I_1 + I_2$ flowing toward b must equal I flowing away from b.

Combined Resistance. Since the sum of two values is larger than either one alone, total current I in the circuit of Fig. 10 is larger than either of the resistor currents. There is only one thing which can make it possible for the source to deliver this larger current—the combined resistance of R_1 and R_2 must be lower than that of either resistor alone. (Remember that lowering the total resistance in a circuit is one way to make the current



FIG. 10. Two resistors in parallel.

increase.) Our conclusion is, then, that the combined resistance of two resistors in parallel is *less* than that of either resistor alone.

Different Resistors in Parallel. When two parallel resistors have different values, the combined resistance will always be less than that of the smaller resistor. This is true because the smaller resistor gets the most current. Any resistor placed in parallel with it will make the total circuit current go up, and this means that <u>the</u> combined resistance of two parallel resistors is less than that of the smaller resistor. See Fig. 11 for two illustrations of this fact.

Identical Resistors in Parallel. When two parallel resistors are alike in ohmic value, the same current must



FIG. 11. When two different resistors are in parallel, the combined resistance is LESS than that of the smaller resistor. Note: Zero is at the extreme right on the scale.

thow through each. The total current is then twice the current through one resistor. For any given voltage value, the current can be doubled only by cutting the resistance value in half. The combined or equivalent resistance of two equal resistors in parallel is thus *exactly half* that of one resistor.

Likewise, the combined resistance of three equal resistors in parallel is exactly one-third that of one resistor. Actual hook-ups of two and three equal resistors in parallel are shown in Fig. 12.



FIG. 12. When two equal resistors are in parallel, the combined resistance is exactly half that of one resistor. When three equal resistors are in parallel, the combined resistance is one-third that of one resistor.



The general rule is: To find the combined resistance of equal resistors in parallel, divide the resistance of one resistor by the number of resistors. Example: Four 40,000-ohm resistors in parallel have a combined resistance of $40,000 \div 4$, which is 10,000 ohms.

Parallel Resistance Formula. The foregoing general information is usually sufficient for a radio man when making resistance measurements in a receiver. There is, however, a fairly simple way of figuring out exactly the combined resistance of two resistors in parallel. Just multiply the two resistance values together, then divide by the sum of the resistance values, and you have the combined resistance. Of course, all values must be in ohms.

As a formula, this method of figuring parallel resistance values appears even simpler. Let R_1 and R_2 represent the values in ohms of the two resistors. The combined resistance R of the two resistors in parallel then is:

$$\mathbf{R} = \frac{\mathbf{R}_1 \times \mathbf{R}_2}{\mathbf{R}_1 + \mathbf{R}_2}$$

In this kind of formula, the horizontal line means that everything above the line is to be divided by everything below the line, after the indicated multiplication and addition have been carried out. A few examples will illustrate this.

Example 1. Find the combined resistance of two 10-ohm resistors in parallel.

SOLUTION: Write the parallel resistance formula, then repeat it with the two known values in place of R_1 and R_2 , as follows:

$$\mathbf{R} = \frac{\mathbf{R}_1 \times \mathbf{R}_2}{\mathbf{R}_1 + \mathbf{R}_2} \qquad \mathbf{R} = \frac{10 \times 10}{10 + 10}$$

Now multiply the two top numbers, add together the two bottom numbers, then divide the results as follows, to get 5 ohms as the combined resistance (exactly half the value of one resistor):

$$R = \frac{100}{20} \qquad R = 5 \text{ ohms}$$

Example 2. Find the combined resistance of 3 ohms and 6 ohms in parallel.

$$R = \frac{3 \times 6}{3 + 6}$$
 $R = \frac{18}{9}$ $R = 2$

The combined resistance is thus 2 ohms. As we expected, this is less than the value of the smaller resistor.

Combining Large and Small Resistors. When two unequal resistors are in parallel and one is more than about 25 times larger than the other, the combined resistance will be very close to the smaller resistance value. In practical work, a radio man wouldn't even bother to figure out the value; he'd say the combined resistance was about the same as the value of the smaller resistor.

Example 1. Find the combined resistance of 1 ohm and 25 ohms.

$$R = \frac{25 \times 1}{25 + 1}$$
 $R = \frac{25}{26}$

R = .96 ohm, which for practical purposes is 1 ohm.

Example 2. Find the combined resistance of 300 ohms and 10,000 ohms.

$$\mathbf{R} = \frac{300 \times 10,000}{10,300} \qquad \mathbf{R} = \frac{3,000,000}{10,300}$$

R = 291.2 ohms, which for practical purposes is 300 ohms.

Three or More In Parallel. When more than two different resistors are in parallel, disregard all resistors which are more than 25 times larger than the smallest. If three resistors are left, combine two of them by means of the formula, then combine the result with the third resistor. By taking resistance values two at a time in this way, you can figure out the combined resistance of any number of resistors in parallel. The combined resistance will always be less than that of the smallest resistor in the group.

Example: Find the combined resistance of 2 ohms, 5 ohms, 10 ohms, 250 ohms and 1 megohm in parallel.

First, we cross out 250 ohms and 1 megohm, since they are more than 25 times larger than 2 ohms. Now we combine 2 ohms with 5 ohms by means of our formula, and get 1.43 ohms. Next, we combine 1.43 ohms with 10 ohms, and get 1.26 ohms as the combined resistance of these five resistors in parallel. Note that this is less than the value of the smallest resistor in the group.



FIG. 13. A series resistor is often used to reduce the voltage for loads in parallel.

Parallel Loads With Series Resistor. In radio, we quite often have a number of loads in parallel, all requiring the same voltage, and a source which has too high a voltage for the loads. A series resistor will reduce the load voltage here, the same as it did for a single load resistor. All you have to do is figure out the combined resistance of all the loads in parallel, then use this value as if it were for a single load resistor.

An arrangement of parallel loads with a series resistor is shown in Fig. 13. The parallel loads require 2 volts, but the source furnishes 3 volts. As a review of the formulas you've already learned, let's figure out what the series resistor value should be. First, we find the combined resistance of the load. The three 30ohm resistors have a combined resistance of 10 ohms, and combining this with the 10-ohm resistor gives 5 ohms as the combined resistance of the group. (We chose easy values purposely here, but you could do this for any values by using the parallel resistance formula.)

Next, we find the current which this combined load resistance will draw through the series resistor. We have a 5-ohm load and it requires 2 volts; $I = E \div R$, so I is $2 \div 5$, which is .4 ampere.

We want to find R for the series resistor, so we bring out another Ohm's Law formula: $R = E \div I$. We've already found that I is .4 ampere. Kirchhoff's Voltage Law tells us that E (the voltage drop across the series resistor) plus 2 volts (the load voltage) must be equal to 3 volts, so E must be 1 volt (3 - 2 = 1). Substituting these values, we find that R is $1 \div .4$, which is 2.5 ohms, the value of the series resistor.

Example. The filament circuit arrangement in Fig. 14 is an excellent example of how a series resistor reduces the voltage for parallel loads in a receiver. The filaments of the five tubes are connected in parallel between terminals a and b, with the connections between ground symbols be-



FIG. 14. Filament circuit arrangement of a 5-tube battery-operated radio set.

ing completed by the metal chassis of the receiver.

Each tube filament requires 2 volts, so the voltage between a and b must be 2 volts at all times. The battery delivers 3 volts when new, but this voltage drops gradually as the battery runs down. This is why a rheostat (variable resistor) is used as a series resistor for reducing the load voltage to 2 volts.

During normal use of this receiver, the owner adjusts the rheostat about once a week by means of a knob at the back of the receiver, in the position shown in Fig. 15. He reduces the ohmic value of the rheostat, so



FIG. 15. Rear view of Belmont Model 524 radio receiver, showing the location of the control knob for the series rheostat which adjusts the filament voltage. The set owner is instructed to turn the rheostat one marked division further each week, which gives approximate correction of filament voltage during normal use.

that the voltage drop across it becomes less and less as the battery runs down. When the battery is down= to 2 volts, the rheostat is down to zero resistance, and further drops in voltage can no longer be corrected. The receiver eventually stops working, and a new A battery must then be installed.

Problem: Load Voltage Varies Too Much. When the load is a resistor, a series resistor does a perfect job of reducing the load voltage to the correct value. When the load is a vacuum tube which acts like a resistance, however, we find that the load voltage just won't stay where it should.

A typical situation of this kind is presented in Fig. 16. The vacuum tube load in this particular example requires 40 volts at all times, and the series resistor is chosen to provide this



FIG. 16. Here's an interesting practical radio problem: The load is a vacuum tube whose resistance varies so much that a voltmeter across the load may read any value between 15 volts and 49 volts when the voltage is supposed to be 40 volts at all times. This lesson shows you how just one extra resistor of the proper value, properly connected into the circuit, will "save the day."

voltage when the tube resistance is at its normal value of 4000 ohms. During normal operation, the tube resistance can drop all the way down to 1000 ohms or go up to 6000 ohms, and this explains why the d.c. voltmeter in Fig. 16 is showing dozens of different readings ranging from 15 volts to 49 volts.



This low-wattage, low-value wire-wound resistor appears exactly like a carbon or metallized resistor. The resistance wire is wound on a core of insulating fiber, wire terminal leads are crimped over each end, and an insulating material similar to bakelite is molded over the entire assembly.

Each change in load resistance changes the total circuit resistance and the circuit current flowing through the series resistor, thus changing the amount of voltage which the series resistor leaves for the load. What's to be done about it?

Clearly, our problem is to make the load voltage independent of the load resistance. We can do this by placing across the load a resistor having considerably lower resistance than the load, as shown in Fig. 17. (This resistor is usually called a *shunt resistor*, because *shunt* and *parallel* have the same meaning in radio.)

After the shunt resistor is connected, we must change the series resistor value so the load still gets 40 volts initially. Now the voltmeter just barely flickers around 40 volts when the tube resistance varies as it did before. For all practical radio purcurrent and all the circuit voltages likewise stay about the same. Thus, the shunt resistor does a fine job of steadying the load voltage.

A shunt resistor is sometimes called a *bleeder* resistor, because it draws or "bleeds" from the source a certain amount of current which flows through the series resistor but never gets a chance to do useful work in the load.

Voltage Regulation. When the load voltage varies excessively with load resistance, radio men say that the circuit has *POOR voltage regulation*. When the load voltage stays essentially the same despite variations in load resistance, the circuit has



FIG. 17. With a suitable shunt resistor value in our circuit, the load voltage can vary only over the limited range of 35 volts to 41 volts, which is hardly enough variation to worry a practical radio man.

poses, we have a perfect solution to our problem. Let's see now why this single extra resistor has such an effect on the behavior of the circuit.

It is the combined resistance of shunt and load which determines how much voltage drop there is across the load. If we use a shunt resistor having much lower resistance than the load, the combined resistance will be just a little smaller than the shunt resistance. Furthermore, variations in load resistance will have little effect on the combined resistance because even the lowest load resistance value is still many times larger than the shunt value.

With the combined resistance staying essentially constant, the circuit GOOD voltage regulation. A shunt or bleeder resistor across a load thus *improves* the voltage regulation of the circuit.

Review Example

The circuit in Fig. 18, representing the plate supply circuits of three radio tubes in a simple public address amplifier, is an excellent illustration of how a serviceman can use Kirchhoff's Laws to speed up trouble-shooting in radio circuits.

All trouble-shooting is based upon a knowledge of *what should exist* when everything is working properly. Thus, a Radiotrician working on this circuit knows from Kirchhoff's Voltage Law that the d.c. voltage drops in



FIG. 18. Simplified plate supply circuits of a public address amplifier, showing voltage and current values a Radiotrician would measure if these circuits were free of trouble.

each complete plate circuit should add up to the source voltage of 270 volts. Voltage values which he might measure during normal operation are indicated in the diagram in Fig. 18, so let's check them.

In tube circuit VT_1 , we add the tube drop of 180 volts to the 90 volts across R_1 , and get just what the law says we should—270 volts. Tube circuit VT_2 gives exactly the same result. In tube circuit VT_3 , we add the tube drop of 250 volts to the 20 volts across R_3 , and again get exactly 270 volts.

Normal circuit current measurements are also indicated in Fig. 18, and can be checked quickly against Kirchhoff's Current Law. Thus, adding .004 amp., .004 amp., and .03 amp. gives .038 amp., which is the source current.

Here are two examples of how a Radiotrician would use his knowledge of radio laws to help locate trouble in this circuit.

Example 1. Instead of the voltage and current values shown in Fig. 18, suppose there is no voltage across resistor R_1 , and tube VT_1 has the full source voltage of 270 volts. In addition, the plate current for tube VT_1 is much higher than the expected value of .004 ampere. What is wrong?

No voltage across R_1 can mean that R_1 is shorted and current is flowing around it through the shorting path, or can mean that no current is flowing through the entire circuit which includes R_1 . We know that plate current is flowing through the circuit, so R_1 must be shorted. Full source voltage across tube VT_1 confirms the fact that there is no series resistance in the circuit, so the Radiotrician investigates R_1 to see if it is defective or if its leads are touching each other.

Example 2. Measurements indicate the same voltage situation as for Example 1, but this time there is no plate current.

Absence of current means that either R_1 or the tube is open. (When a tube does not pass current, it actejust like an open resistance.) Full voltage across the tube proves that continuous paths exist from the tube terminals to the voltage source, so Rcannot be open. This means that the tube is guilty this time. Whenever a break occurs in a complete circuit such as at the tube here, full source voltage appears across the break, just as if your meter were connected directly to the source terminals.

Controlling Load Voltage with Potentiometers

In radio, it is often necessary to have a circuit arrangement which makes it possible to change the voltage of a load by turning a knob. You have already seen one way of doing this, by means of a rheostat in series with the load. A rheostat can reduce the load voltage only a certain amount, however—never all the way down to zero.

When complete control of load voltage from the maximum value to zero is required, we use a special type of variable resistor known as a potentiometer (pronounced *poe-TEN-shiom-eh-ter*). The basic circuit is shown in Fig. 19.

As you can see, a potentiometer is simply a rheostat having three terminals instead of two. The entire resistance element of the potentiometer is connected across the source. The load is connected between the movable contact (P) and one of the end terminals of the resistance element, and hence the load can be connected across any desired portion of the resistance element by changing the position of the movable contact.

When P is set at X, the load gets the full source voltage (lower left in Fig. 19).

When P is moved downwards, such as to the position shown at the top in Fig. 19, the upper potentiometer section (R_1) is a series resistance, and the lower potentiometer section (R_2) is a shunt resistance connected directly across the load. The voltage drop across R_1 now cuts down the amount of voltage available to the load. Note that both the load current and the current drawn by R_2 flow through series resistance section R_1 and produce this voltage drop.

Moving P farther downward reduces the load voltage still more. When P is moved all the way down to Y, as at the lower right in Fig. 19, the load voltage is zero.



FIG. 19. Basic circuit using a potentiometer

A potentiometer is often thought of as a voltage divider, because it divides the source voltage in the desired proportion between two parts of a circuit (in this case, it divides the voltage between R_1 and the load- R_2 parallel combination).

When the total resistance of the potentiometer is much lower than the load resistance, the current drawn from the source depends chiefly on the resistance value of the potentiometer. The source current then stays essentially the same regardless of the position of the movable contact; this is a highly desirable condition in some radio circuits.

Wattage Ratings of Resistors

If you glance through any catalog or listing of resistors, you will see that in addition to the ohmic value, each resistor has a *wattage rating*. This rating tells a radio man whether or not a resistor can "take it" in a particular circuit without getting too hot or burning out.

The rate at which a resistor produces heat is its *power* in *watts*. This value must for safety be lower than the wattage rating, because the wattage rating represents the maximum rate at which the resistor can safely produce heat. Since practical radio men work with resistors just about every day, the subjects of power, wattage ratings and watts are an important part of our study of resistors.

What is POWER? Let us suppose that we have two quart pans full of water, which we want to boil. If we set one pan on a small electric hot plate and the other on the large superspeed burner of an electric range, it may take the small burner three times as long to boil the water. Both burners accomplish the same result, however, for both have to supply the same total amount of heat to the quart of water.

The large burner accomplishes its work at three times the *rate* of the small burner. The *rate of doing work* —in this case, of producing heat—is called power.

This definition of power holds true for radio, too, because a resistor produces heat at a definite rate when current flows through it. Before we consider power problems in resistors, however, let us first get acquainted with the practical *units* used by radio men for specifying amounts of power.

Units of Power. In radio and electricity, power is measured in *watts* or in *kilowatts*. One kilowatt is equal to 1000 watts, hence one watt is onethousandth of a kilowatt. Small radio sets draw about 40 watts, which is the same as the power drawn by a 40-watt soldering iron or a 40-watt electric lamp bulb. Large sets draw as much as 150 watts.

In radio, we occasionally encounter a smaller unit of power called the *milliwatt*. This is equal to one-thousandth of a watt, and is used chiefly in designating the strength of audio signals.

Figuring Power. To find the amount of power which a voltage



This wire-wound resistor looks much like a carbon or metallized resistor. It is made with various ohmic values, ranging from 5 ohms to 70,000 ohms, and in two different wattage ratings—5 watts or 10 watts. A red dot on the resistor changes to brown when the resistor is overloaded (when the wattage rating is exceeded). The dot changes back to red when the overload is corrected.

source is delivering to a circuit, all you need to do is multiply the amount of voltage by the amount of current. The voltage should be in volts and the current in amperes; the result will then be the electrical power in watts which is being delivered to the circuit by the voltage source. Thus, if a 6-volt battery is furnishing 2 amperes, the power being delivered is 12 watts (6×2) = 12).

To find the amount of power which is producing heat in a resistance, simply multiply the current flowing through the resistance by the voltage drop across the resistance. Voltage must be in volts and current in amperes. The result will then be the power in watts which is producing heat. Thus, if a current of 4 amperes produces a voltage drop of 5 volts across a resistor, the power in the resistor is 20 watts $(5 \times 4 = 20)$.

Power Formula. In radio, we let the capital letter P stand for power. The simple formula which expresses what has just been said about figuring power is:

$\mathbf{P} = \mathbf{E} \times \mathbf{I}$

Remember that P stands for the power in watts, E stands for the voltage in volts, and I stands for the current in amperes.

This power formula applies to all direct current circuits, and to all a.c. circuits containing only resistance.

Radio men use the power formula chiefly to figure how much power a resistor is handling. A given resistor can handle only a certain amount of power without getting too hot and destroying itself, because all of the power supplied to a resistor is used to produce heat. This is the reason why resistors have wattage ratings in addition to their resistance values.

Wattage Ratings of Resistors. The wattage rating of a resistor is the maximum amount of power in watts which the resistor can handle without becoming excessively hot. Naturally, a resistor can safely handle any power value which is *less* than its wattage rating.

The wattage rating depends essentially on the physical size of a resistor (length, thickness and surface area), and is entirely independent of the ohmic value. In plain language, the wattage rating depends upon how fast the resistor can get rid of the heat produced by its resistance element.

The larger the physical size of a resistor, the more heat it can withstand and the higher will be its wattage rating. A 10-watt resistor will thus be larger in dimensions than a 1-watt resistor.

As a matter of fact, in a radio store

you could get a resistor with a suitable wattage rating by holding up the old resistor and saying, "I want a 500-ohm resistor which is this big or bigger." When ordering a resistor from a mail order radio supply firm, however, you would have to specify the wattage rating as well as the ohmic value.

Most of the resistors which you will encounter in radio circuits have low wattage ratings, such as $\frac{1}{4}$, $\frac{1}{2}$, 1 or 2 watts. A $\frac{1}{4}$ -watt resistor is less than



FIG. 20. Examples of carbon resistors having different wattage ratings.

an inch long, and is scarcely any thicker than a piece of insulated hookup wire. Figure 20 shows how the physical dimensions of resistors vary with the wattage rating.

Margin of Safety. It is good practice to provide a margin of safety by using a resistor having a wattage rating which is about twice the actual power it must handle. This prevents resistor failure even though the power should for some reason get considerably higher than the normal value.



Another reason for providing a margin of safety is that wattage ratings are for resistors mounted in free air. When resistors are mounted in enclosed places, such as under a shallow chassis where there is little circulation of air, the heat-dissipating ability is greatly reduced.

Examples. If a resistor is to handle 4 watts, choose a 10-watt resistor for safety. If it is to handle .4 or .6 watt, use a 1-watt resistor. If it is to handle 1 watt, use a 2-watt resistor. As a general rule, try to use a wattage rating which is about twice the actual power being handled.

Rheostat Wattage Ratings. Wattage ratings of variable resistance units, such as rheostats and potentiometers, apply to the *entire resistance element*. When the position of the movable contact is such that only a part of the resistance element is in the circuit, the wattage rating is *proportionately reduced*.

Thus, if the movable contact of a rheostat is at the mid position, the unit can safely handle only half of the wattage rating. If the movable contact is set so only about one-eighth of the total resistance is in the circuit, the unit can safely handle only oneeighth of the wattage rating.

Potentiometers and rheostats invariably burn out near a terminal, because only a small portion of the resistance element is then in the circuit, and the useful wattage rating is correspondingly small.

A burned-out potentiometer is shown in Fig. 21. When the contact arm was set near the right-hand terminal, the wire which forms the resistance element became so hot that it melted and the surrounding fiber of the unit became badly charred.

If you encounter a receiver in which a certain resistor, rheostat or potentiometer becomes too hot and perhaps even burns out frequently, yet no other parts in the receiver are defective and causing excessive current to flow, you should replace the overheated unit with another having the same ohmic value but a *higher* wattage rating.

Wattage Ratings of Parallel Resistors. When two or more resistors which have the same ohmic values and the same wattage ratings are connected in parallel, the wattage rating of the combination will be the sum of the ratings of the individual resistors.



FIG. 21. Burned-out wire-wound potentiometer.

Thus, if two 100-ohm, 1-watt resistors are connected in parallel, the combined wattage rating is 1 + 1, or 2 watts. (The combined resistance is determined in the usual way by the parallel resistance formula, and is 50 ohms in this case.) If four 40,000-ohm, 3-watt resistors are connected in parallel, the combined wattage rating is 3 + 3 + 3+ 3, which is 12 watts.

When resistors having different wattage ratings and different ohmic values are connected in parallel or in series, the power must be figured for each resistor separately, to make sure that no resistor will be overloaded.



Other Power Formulas. Since the *resistance* value affects the amount of current which flows through a resistor, resistance also affects the amount of power being handled by a resistor. If we know the resistance value and either the voltage or current value, we can figure the power. Here are the formulas:

When the current and resistance are known, use this formula: $P = I \times I$ $\times R$. This is usually written as P = $I^2 \times R$ or simply as $P = I^2R$ (pronounced P equals I squared R). The small numeral 2 at the upper right of a letter means that the value is to be multiplied by itself. Thus, if the current through a resistor is .5 ampere and the resistance is 40 ohms, the power will be $.5 \times .5 \times 40$, which is $.25 \times 40$, or 10 watts.

When the voltage and resistance are known, use this formula: $P = E \times E$ $\Rightarrow R$, which is usually written as P = E^2/R (pronounced *P* equals *E* squared over *R*). If the voltage drop across a resistor is 20 volts and the resistance is 80 ohms, the power will be 20×20 $\Rightarrow 80$, which is $400 \div 80$, or 5 watts.

Important: You are not expected to memorize any of the formulas in this lesson. You can always refer to this lesson when you have need for a particular formula. Eventually, after using a formula a few times, you'll find that you know it without ever having tried to remember it.

How Resistors Are Made

Many Kinds of Construction. Resistors used in radio equipment not only have different ohmic values and different wattage ratings, but they also differ greatly in actual construction. Thus, fixed resistors can have wirewound, carbon or metallized construction.

Some types of construction give greater accuracy than others; some are cheaper than others when mass production methods are used; some provide the high wattage ratings required for resistors which must withstand a lot of heat; some construction methods give extremely low or extremely high ohmic values.

We will first consider the materials from which resistors are made, then study the three types of construction methods used for resistors, and finally get acquainted with typical resistors used in radio.

What Resistors Are Made Of

If we took a piece of ordinary No. 18 lamp cord or bell wire long enough to reach all the way across the country from New York to San Francisco, and measured the total *resistance* of this wire, the result would be less than 100,000 ohms! To get 1 megohm of resistance, a value commonly used in radio circuits, we would have to run this copper wire all the way around the world and still add on about 5000 miles of wire.

You say, "Use smaller wire, because it has much higher resistance per foot of length. Some copper wire is made as fine as human hair."

All right, suppose we did use copper wire the size of a human hair. It would still take 190 miles of the wire to get a megohm of resistance, and that's a lot of wire to put into a radio part.

All this leads us to the conclusion that the resistance of copper is too low for use in resistors. This low-resistance characteristic of copper makes it ideal, however, for the hook-up wire used in connecting radio parts together, and for constructing radio parts which should have low resistance, like coils and transformers.

What we need for practical radio resistors are materials having much higher resistance than copper. Two such materials are *nichrome* (used in wire-wound resistor construction) and *carbon* (used in carbon resistor construction). In addition, there is a *metallized* resistor construction which is often used in place of carbon but has slightly different characteristics.

Resistivity. The easiest way to compare the resistances of different materials is by considering pieces of wire which are all the same length and thickness, each made from a different material. The resistance in ohms of a specified length of wire is technically known as the *resistivity* of the material.

If, for convenience, we choose a length and thickness which will make the copper sample have a resistance of exactly 1 ohm, then similar samples of other common resistance materials will have approximately the ohmic values given in Table I. These values vary considerably in some cases, depending upon the purity of the metal and the manner in which it is made into wire.

Length. Increasing the length of a piece of resistance material without changing its thickness is just like placing resistors in series to get more resistance. Increasing the length IN-CREASES the amount of resistance.

Technically speaking, the resistance of a conductor is *directly proportional* to its length. In other words, 2 feet of wire will have two times the resistance of a 1-foot length; 7 feet will have seven times the resistance of a 1-foot length; one inch of wire will have onetwelfth the resistance of a 1-foot length.

Thickness. Increasing the thickness (cross-sectional area) of a piece of resistance material is just like placing resistors in parallel. As you will recall, resistors in parallel have less resistance than either resistor alone. Increasing the thickness thus *DE*-*CREASES* the resistance. Actually, doubling the cross-sectional arca cuts the resistance exactly in half.

Reducing the thickness of a piece of resistance material *INCREASES* the amount of resistance, because this is

Silver	.9 ohm
Copper	l ohm
Aluminum	1.7 ohms
Iron, soft	6 ohms
Manganin	25 ohms
Constantan	29 ohms
Nichrome	60 ohms
Carbon	2000 ohms

Table 1. Resistances of wire samples which all have the same length and thickness but are made from different materials. The last five materials are used in resistors and are taken up later in this lesson. Note that carbon has 2000 times as much resistance as copper.

like slicing a resistor lengthwise and using only part of it. Cutting the cross-sectional area in half doubles the resistance.

Resistor Construction

Wire-wound Construction. Resistors made from metal wire are known as wire-wound resistors. The metal wire used most often in resistor construction is an alloy called nichrome, made chiefly from nickel and chromium. As indicated in Table I, nichrome has about 60 times the resistance of copper. This means that the resistivity of nichrome is about 60 times that of copper.

The nichrome wire is wound on a flat strip of fiber, on a rod of porcelain, or on some other stiff insulating material. The turns may be spaced out so they do not touch each other, or the wire may have enamel insulation. When higher-wattage wire-wound resistors are required, they are wound on large cement or baked clay forms. The wire used is made to have the largest possible cross-sectional area, so there is a large surface area to conduct the heat to the form and to the surrounding air.

If a resistor of low olmic value is required, a few turns of fine wire wound on the form would give the desired resistance, but this short wire would not get rid of its heat quickly enough to give the desired wattage rating.

Thicker and longer wire having the same resistance value is better, because it will get rid of heat faster. Increasing length and thickness both increase the surface area which radiates heat, so the resistor can then radiate more heat safely. This makes possible a higher wattage rating.

As you know, heat is produced whenever current flows through a resistance. The larger the current, the more heat we get. Nichrome wire can get red hot without breaking or melting; in fact the material used for the heating elements of electric stoves, toasters, electric heaters and soldering irons is usually nichrome.

But nichrome is by no means the complete solution to our resistor-making problem. We would still need a little over three miles of the smallest size nichrome wire to get a million ohms of resistance, and the cost of that much wire would be too high for practical purposes.

Nichrome wire is ordinarily used only for fairly low resistance values (below about 50,000 ohms for both fixed resistors and rheostats or potentiometers). Even then, wire-wound resistors made from nichrome (or similar metal alloys sold under various other trade names) are used chiefly when the resistor must carry considerable current, when the ohmic value of the resistor must be highly accurate, or when the ohmic value must stay the same despite changes in temperature.

Carbon Construction. This is the construction method which makes possible the tiny high-value fixed resistors, rheostats and potentiometers found in modern radio sets.

By itself, carbon has a resistance ranging from 500 to 2500 times the resistance of copper. By mixing carbon particles with certain cements or clay products, however, we can increase the resistance to more than a million times that of a corresponding amount of copper. Cements or binders serve the added purposes of giving needed "backbone" or strength to carbon, and keeping the carbon particles in position so the resistance stays fairly constant. Carbon is really a good all-around resistance material for general use.

By varying the proportion of carbon to binder material and forming the mixture into rod-shaped pieces, fixed carbon resistors having the same physical size (length and thickness) can be made in any desired value from about 50 ohms to about 15 megohms.

A carbon resistor scarcely half an inch long can have a resistance of a million ohms. Compare this with the miles of metal wire which would be needed to get such a high resistance.

The ends of each carbon resistance element are copper-plated so terminal leads can be soldered to them. The completed unit is then painted with various colors of enamel so as to specify its ohmic value according to the standard resistor color code (explained elsewhere in the Course).

Sometimes carbon resistors are placed in tubes of insulating material or coated with an insulating material. More often, the insulating material is molded over the carbon resistance element. Color code markings are then applied over the insulation. Insula-



tion prevents short-circuits (accidental short-cut paths for current) when resistors touch each other or other parts in a crowded chassis.

The chief reason for making carbon resistors in larger physical sizes is to secure higher wattage ratings. A larger resistor has more surface area to radiate the heat which is generated in the resistor, and can therefore handle more electrical power without becoming too hot. Three watts is about the highest rating in which carbon resistors are ordinarily made, however, for carbon-cement mixtures cannot withstand much heat.

Carbon resistors are the lowestpriced of all resistors, and for this reason are widely used in ordinary radio equipment.

Carbon rheostats and potentiometers are commercially available in values from about 500 ohms to about 10 megohms. Thus, values below 500 ohms are ordinarily obtainable only as wire-wound units, and high ohmic values are available only as carbon units.

Metallized Construction. In this process of making resistance elements for resistors, a liquid solution of certain metals is produced by a secret chemical process. The backbone of the resistor, usually a glass or porcelain rod or tube, is coated with this liquid and heated. This evaporates the liquid, leaving on the glass a hard, thin metallic coating which serves as the resistance element. By varying the strength of the metallic solution and by varying the amount of solution sprayed on the backbone, any desired resistance value between about 250 ohms and 20,000,000 ohms can be obtained.

How Temperature Affects Resistance. Practically all metals *increase* in resistance a small amount when heated. For example, a piece of *nich*rome wire having a resistance of 1 ohm when in ice water would have a resistance of about 1.04 ohms when placed in boiling water (this is a temperature rise of 180° Fahrenheit). A piece of *copper* wire having a resistance of 1 ohm at freezing temperature would increase much more, to about 1.39 ohms, when placed in boiling water.

Constantan and Manganin are special metal alloys which are little affected by temperature. Thus, a 1-ohm piece of Manganin would increase only to 1.001 ohms when transferred from freezing water to boiling water. These and other similar alloys are used chiefly for making precision resistors used in meters and in laboratory equipment requiring highly accurate and stable resistance values.

Carbon actually drops in resistance when temperature is increased, and this drop can be quite large when carbon resistors heat up during use.

When a material having a resistance of one ohm is heated or cooled one degree, the resulting CHANGE IN OHMIC VALUE is known as the temperature coefficient of resistance

Metals are said to have a *positive* temperature coefficient of resistance, because they *increase* in ohmic value when temperature is increased. Carbon, on the other hand, has a *negative* temperature coefficient of resistance, because it *decreases* in ohmic value when heated.

Pilot lamps and tube filaments are perhaps the most striking examples of how the resistance of a metal increases with temperature. In these parts, the change in temperature from a cold filament to a red or white-hot filament may be several thousand degrees.

The hot resistance of a tungsten filament may be as much as ten or fifteen times its cold resistance. Thus, a particular pilot lamp might have a resistance of 4 ohms if measured with an ohmmeter when cold, and a resistance of 40 ohms when hot. (The hot resistance cannot be measured with an ohmmeter, but can be figured out with Ohm's Law if the pilot lamp voltage and current are known or are measured.)

Tube filaments do not increase quite as much when hot, because these filaments do not get as hot as those in pilot lamps. Just remember the general fact-that a tube or lamp filament has a higher resistance when hot This will explain than when cold. seemingly erroneous results which might be obtained when using lamps as resistances in experimental circuits. It also explains why tube filaments often burn out right after a receiver is turned on. The cold tube filaments have a low resistance, hence an excessively large initial current flows when the set is first turned on.

Lamps with carbon filaments are now quite rare, but it is nevertheless worth pointing out that these lamps have a *much lower* resistance when hot than when cold.

Iron or nickel are resistance materials which increase in resistance even more rapidly than nichrome when heated. This characteristic is utilized to advantage in the special resistance tubes which serve as filament voltage-dropping resistors in some universal a.c.-d.c. receivers. These resistance tubes are so designed that when normal current flows through the resistance wire and produces normal heat, all tube filaments in the receiver get correct voltages.

If the line voltage goes above normal, however, the increased current through the resistance tube makes its resistance increase, so the resistance tube takes most of the increase in line voltage. As a result, the tubes still receive normal voltage, and are **thus** protected from burn-outs.

In the same way, when the line voltage drops lower than normal, the resistance of this resistance tube drops, the voltage drop across the tube becomes lower, and the tube filaments still get essentially the same voltage even though the line voltage has dropped.

Tolerance of Resistors. In resistors, just as in other manufactured parts, there is no such thing as a perfectly accurate unit. Accuracy costs money, because it means additional testing and more careful adjustment of resistor values. A 100-ohm, 1 watt resistor which is accurate to within 20% may cost only four cents, whereas the same resistor with an accuracy of 1% may cost fifty cents, and with .1% accuracy may cost \$2.

The per cent variation from the specified ohmic value is called the tolerance of a resistor. A tolerance of 20%, sometimes expressed as $\pm 20\%$ (plus or minus twenty per cent), means that any value in the range from 20% under to 20% over the specified resistor value is tolerated (passed) during manufacture.

Thus, a 100-ohm resistor with $\pm 20\%$ tolerance may actually be anywhere between 80 ohms and 120 ohms. (20% is .2; .2 \times 100 = 20; 100 - 20 = 80; 100 + 20 = 120). A 100-ohm resistor with 5% tolerance may be anywhere between 95 and 105 ohms.

A 500,000-ohm resistor with 20% tolerance can be anywhere between 400,000 and 600,000 ohms.

A resistor tolerance or accuracy of 10% or even 20% is entirely satisfactory for general use in radio circuits, because resistance values are not critical in most circuits. Only in measuring instruments and in certain special radio circuits is it necessary to use more accurate resistors. Normally, carbon resistors are made only to an accuracy of 20%. For special applications, however, carbon resistors are available with a tolerance of 10% or 5%, depending upon circuit requirements.

Metallized resistors have a normal accuracy of 10%, but here again, more accurate units are available at increased cost. Even 2% accuracy is obtainable, because these resistors are more stable (less affected by temperature changes) than carbon resistors.

Wire-wound resistors can be made as accurate as .1%, but 5% is the normal tolerance for general-purpose wire-wound resistors. Greater permanent accuracy and greater powerhandling ability are the two main features of wire-wound resistors.

The tolerance encountered in actual resistors makes it unnecessary to figure resistance values with extreme accuracy. This is why practical radio men think nothing of using a 1000ohm resistor when their figuring indicates that 900 ohms is needed.

Types of Resistors Used in Radio Work

FIXED RESISTOR. A radio part having an essentially constant ohmic value, used to introduce a definite resistance value permanently into a circuit. Fixed resistors have two terminals, leads or lugs, one at each end of the resistance element.

Three types of fixed resistor construction are found in radio: 1. Carbon Resistors; 2. Metallized Resistors; 3. Wire-wound Resistors.

• 1. Carbon Resistor. A fixed resistor having as its resistance element



a mixture of carbon and a claylike material which has been formed into a rod and baked until hard. Carbon resistors are made in various wattage ratings from $\frac{1}{10}$ watt up to 3 watts. They are the lowest-priced of all resistors. Standard tolerance is 20%, but resistors with 10%, 5% or 2% tolerance are available at higher cost. Carbon resistors always decrease-in-ohmic value when heated. Both insulated and non-insulated types are in common use. The wire leads come out either from the ends or sides.

• 2. Metallized Resistor. A fixed resistor having wire leads attached to

the ends of a metallized resistance element, with the entire



unit encased in cement, bakelite or other insulating material. They have essentially the same appearance, wattage ratings and ohmic values as carbon resistors, and can ordinarily be used interchangeably in radio circuits.

◆ 3. Wire-Wound Resistor. A fixed resistor made of nichrome wire wound on a strip or rod of insulating material. The two ends of the resistance wire are soldered, welded, riveted or clamped to terminal lugs or leads.

Flat wire-wound resistors with exposed wire usually have ohmic values



ranging from 1 ohm to 1000 ohms, and a wattage rating of only $\frac{1}{2}$ watt. These resistors are therefore suitable only for circuits where both resistance and power-handling requirements are low.

Flat wire-wound resistors wrapped in heavy fiber paper, then encased in a

piece of sheet steel, are sold under the trade name



"Candohm." The metal shield protects the resistance wire from damage and helps to get rid of heat.



In wirewound resistors which must handle a great deal of power, the resistance wire is wound on a porcelain

tube and is covered with a baked-on layer of vitreous (glass-like) porcelain enamel or a special heat-resisting cement.

TAPPED RESISTOR. A fixed wire-wound resistor having one or more extra permanent terminals or taps along the length of the resistance element, and used chiefly as a voltage divider in the power pack of a radio set. Some tapped resistors have the



resistance wire completely exposed, but more often the wire is protected with a vitreous enamel coating, a combination bakelite and steel housing, or a fiber and steel wrapping. Carbon and metallized resistors are not made with taps, but two or more of these resistors can be used in series to get taps.

ADJUSTABLE RESISTOR. A fixed wire-wound resistor having one or more semi-movable extra terminals or taps which can be clamped in position at desired points along the length of the resistor. A semi-adjustable rheostat is obtained by using one sliding terminal and one end terminal. A potentiometer is obtained by using one

sliding terminal and both end terminals. A screwdriver or wrench is required to change the resistance value. Adjustable resistors are used chiefly as voltage dividers in power packs.



RHEOSTAT. A resistor which can be changed in ohmic value by rotating a knob. In wire-wound rheostats, bare nichrome resistance wire is wound around a strip of flexible insulating material. The strip is then bent into a ring and mounted so a contact arm can be rotated over the edges of the turns of wire. One terminal of the rheostat goes to one end of the resistance wire, and the other terminal goes to the movable contact.

When the contact arm is at the fixed terminal, the resistance between the two terminals is zero. When the contact arm is moved away from the fixed terminal, the resistance between the terminals increases. When the contact



SYMBOLS FOR A RHEOSTAT

arm is farthest away from the fixed terminal, the resistance is a maximum.

In one type of carbon rheostat, a shallow groove in a round disc of bakelite is filled with a mixture of carbon and cement. In another type, a washer-shaped disc of paper or the

inside of a paper ring is sprayed with a solution containing carbon. This circle of resistance material is broken at one point to form the resistance element over which the contact slides.

POTENTIOMETER. A rheostat having terminals at both ends of the resistance element, making three terminals in all. Construction is exactly the same as for rheostats; in fact, a potentiometer can be used as a rheostat by leaving one end terminal unconnected. Carbon units have a maxi-



mum wattage rating of one watt or less, hence are used chiefly in low-current signal circuits. Wire-wound units ordinarily have 5-watt ratings in radio work, but larger units are available.

RESISTANCE TUBE. A length of iron, nickel or nichrome wire mounted



SYMBOL FOR RESISTANCE TUBE



in either a glass or

metal radio tube hous-

ing, and used in series

with the filaments of

produced by the tube is dissipated much faster than by an equivalent under-chassis resistor.

Resistance tubes using iron or nickel wire are sometimes known as ballast tubes, because they offset variations in line voltage by changing in resistance when the line voltage changes.

Resistance tubes usually have one or more taps, so pilot lamps can be connected across part of the resistance. The schematic symbol for a resistance tube shows the internal connections between the resistance wire and the base prongs or socket terminals.

LINE CORD RESISTOR. A spirally wound resistance wire covered with asbestos,

r u n n i n g alongside the two regular stranded copper wires of the power



cord in a universal a.c.-d.c. receiver. The cord gets warm when in use, since it must get rid of the same amount of heat as a resistance tube.

SUPPRESSORS. Fixed carbon or wire-wound resistors used in series

with spark plug and distributor cables of automobile ignition systems to suppress interference which might affect auto radio reception. They have special terminals



to permit easy mounting on spark plugs and distributor.

Looking Ahead

In the next lesson, you will learn about coils in much the same way that you have just learned about resistors. You will find that coils act like resistors in some ways, but coils act entirely different from resistors in alternating current circuits. The manner in which a coil can change the amount of its opposition instantly when the frequency of the current changes is one of the truly fascinating actions in radio.

Lesson Questions

Be sure to number your Answer Sheet 5FR-4.

Place your Student Number on every Answer Sheet.

Send in your answers for this lesson immediately after you finish them, as instructed in the Study Schedule. Do this, and get the greatest possible benefit from our speedy personal grading service.

- 1. What does each of the following represent:
 - (a) E
 - (b) I
 - (c) R
 - (d) ω
 - (e) MEG.
- 2. Give the formula you should use to find the amount of current flowing in a circuit, if you know the source voltage and the total resistance of the circuit.
- 3. In the circuit at the right, the load voltage E is 12 volts, and the load current I is 3 amperes. The formula to use is $R = E \div I$. Is the load resistance value: 4 ohms; 36 ohms; or $\frac{1}{4}$ ohm?



5. In the circuit at the right, the source voltage is 6 volts, and the load voltage is 4 volts. How much is the voltage drop across series resistor R?



I . 3 AMPERES

12 VOLTS

- 6. Currents of 3 amperes, 7 amperes, and 2 amperes are flowing toward a terminal. How many amperes of current are flowing away from that terminal?
- 7. If a 5 ohm, 10 ohm, and another 10 ohm resistor are connected in parallel will the total resistance be: 25 ohms; 20 ohms; or less than 5 ohms?
- 8. Write one of the formulas for finding the power dissipated in a resistor.
- 9. About how much higher than the actual power should the wattage rating of a resistor be to give a margin of safety?
- 10. When the filament of a vacuum tube is heated does the filament resistance: increase; decrease; or remain the same?



A PLAN FOR YOUR FUTURE

In a radio interview a few minutes after a championship heavyweight boxing match, one of the fighters stated his plans for the future in just fifteen words:

> "I'm going to get myself in shape, fight my own fights, and listen to nobody!"

You can use these dynamite-packed words as your plan for the future, too. Here's the way:

"GET MYSELF IN SHAPE." You're doing this right now, because the N.R.I. Course gets you in shape for a career in radio. But remember that it takes the complete N.R.I. Course, with all its associated practical work, to get you completely in shape.

"FIGHT MY OWN FIGHTS." In real life, the only person who can bring you success is YOU yourself. Expecting somebody else to do your work and fight for your success is just wishful thinking.

"LISTEN TO NOBODY." Even friends and relatives will at times ridicule your studies—they can't help it, because seeing you get ahead makes them feel uncomfortable about their own laziness. So, remember human nature, and don't give anyone a chance to discourage you.

J. E. SMITH.



STUDY SCHEDULE NO. 6

This schedule will help you to master the sixth part of your N.R.I. Course. For each step, read the specified pages first at your usual reading-speed, then reread slowly one-or more times. Finish with one quick reading, then answer the Lesson Questions specified for that step.

- 1. Quickly Review Pages 12 to 18 of Lesson 1FR-3. Start with "Radio Uses for Magnetism" on page 12, and read to the bottom of page 18 just once.

- □ 4. Using Coils to Produce Voltage......Pages 8-14 Step by step you learn the meaning of the basic coil rule that the induced voltage depends upon how fast the flux linkages are changing. Finally, you fix this knowledge in your mind by considering three examples of voltage-producing coils: Dynamic microphone, magnetic phono pick-up and transformer. Answer Lesson Questions 4, 5, 6, and 7.
- □ 6. Simple Coil-Resistor Circuit......Pages 24-28 Here's where you learn how to combine voltages correctly in a.e. circuits containing a coil. Phase relationships and uses for vectors are explained simply from a practical standpoint. Answer Lesson Question 10.
- **7.** Mail Your Answers for Lesson 6FR-3 to N.R.I. for Grading.

9. Start Studying the Next Lesson, on Condensers.

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World Radio History

RADIO COILS AND HOW THEY WORK

Coil Fundamentals

Radio Uses for Coils. Coils have many important jobs in radio receivers and transmitters. Working alone or in partnership with condensers and resistors, coils make it possible to tune in stations, to remove hum from power supply circuits, to keep signals in proper paths, to change a.c. voltages to higher or lower values, to transfer signals from one circuit to another without wire connections, and to do a host of other equally important jobs, all of which we will take up in this Course.

Definition of a Coil. In its simplest form, a *coil* is nothing more than

one or more turns or loops of wire, usually wound in a circular or helical shape. More often, however, a radio coil also has certain accessories, such as a coil form, an iron core, or a metal housing or shield.

Basic Action. Coils produce magnetic lines of force, known also as *magnetic flux*. No matter what kind of coil we have—large or small, thick or thin, wound around air or iron—it produces magnetic flux whenever we send current through the coil.

Many different types and sizes of coils are used in radio, in order to get a particular desired amount of this



Courtesy Western Miestrie Co.

In this Western Electric magnetic-tape sound recorder, coils have many important jobs. A coil in the dynamic microphone produces the audio signal. An a. f. input transformer inside the cabinet transfers the signal to the first a.f. amplifier stage. Other transformers transfer the signal to succeeding amplifier stages to boost its strength, then feed it to a coil which magnetizes a moving steel tape. Sounds are thus recorded on the long tape. A flip of a switch reverses the process, enabling the speaker to hear her own voice from the loudspeaker, which has two more coils. Still another coil demagnetizes the tape to erase previous recordings. magnetic flux in a particular and definite location inside or near the coil. This flux makes it possible for a coil to offer more opposition to the



FIG. 1. Review of the basic principle applying to coils—that magnetic lines of force are produced whenever electrons flow through a wire or other object.

flow of alternating current than direct current, to transfer energy to another circuit without connecting wires, to produce mechanical motion, to convert mechanical motion into a correspondingly varying voltage, and do all the other jobs which are described in connection with coils in this and later lessons.

Fundamental Facts About Coils. You have already learned that whenever we send current through a straight piece of wire, magnetic lines of force are produced around the wire. These lines of force encircle the wire in much the manner shown in Fig. 1A. They are distributed along the entire length of the wire, because the electrons are distributed throughout the wire and each electron has its own circular magnetic field. The larger the current (the greater the number of electrons) we send through the wire, the more lines of force there will be-just as we might naturally expect. Furthermore, reversing the direction of the current will reverse the direction of the magnetic lines of force.

If we bend the wire into a loop or single-turn coil as shown in Fig. 1B, the lines of force all pass through the center of the loop in the same direction and give a concentration of magnetic flux there. The greater the current, the more flux we have passing through the coil.

If we add more turns of wire as in Fig. 2, still more flux is concentrated inside the coil. Each current-carrying turn of wire contributes its share of flux to the total amount, so the more turns of wire we have in a coil, the more flux we get from a given current.







The number of turns is, therefore, just as important a factor as current in determining how much flux a coil will produce. Many of these facts you will recognize as a review of what you have already studied about magnetic lines of force.

MAGNETIC CIRCUITS

The magnetic lines of force which are produced by a coil can exist only as complete loops having no ends, passing through and around the coil turns which produced the lines. Thus, the lines of force going out of the coil at the left and into it at the right in Fig. 2 are each portions of complete loops like those shown above and below the coil. In many cases, it would be impractical to show complete loops on diagrams, so for convenience we show only those portions of the loops in which we are most interested.

Each coil may have thousands of such loop paths, all passing through part or all of the coil and going out in all directions from the ends of the coil. Together, the paths of these magnetic lines of force make up the magnetic circuit of the coil.

The material inside a coil is called the *core* of the coil. Sometimes the core material is continued around the outside of the coil, and then forms the chief magnetic circuit of the coil.

When the path for the magnetic lines is entirely through air, as it is in Fig. 2, we have an *air-core coil*.

When a coil of wire is wound or placed around an iron or steel ring, as shown in Fig. 3, magnetic lines of force are produced in this core by the current flow in the coil turns. As we will shortly see, the iron core is a better magnetic circuit than air, and a greater number and concentration of flux lines exists. This is called an *iron-core*



FIG. 3. A coil wound on an iron ring produces more flux and a greater concentration of flux than a similar coil having only air for a core.

coil, and you will often meet it in radio work. The iron core will usually have some other shape than a ring, but there will usually be a continuous



Courtery Western Blostrie Co.

The direction-finding loop antenna of this 30ton Douglas DC-4 transport plane, being inspected here by a United Air Lines communication engineer, is nothing more than a coil enclosed in a tubular aluminum hoop. It is mounted inside the nose of the plane, but the magnetic fields of radio waves from beacon stations go right through the non-magnetic metal covering of the plane and through the loop housing.

path through iron for the magnetic lines of force.

Three Magnetic Terms. The magnetic circuit of a coil is just as important to us as the electric circuit in which the coil is connected. In fact, magnetic circuits are similar to electric circuits in many ways, and have terms which correspond to the voltage, resistance and current of an electric circuit. The terms are magnetomotive force, reluctance and flux, and they have the same relationship to each other as that expressed by Ohm's Law for voltage, resistance and current in an electric circuit.

1. Magnetomotive Force. Just as we have an electromotive force or voltage which forces current around an electric circuit, so do we have in every current-carrying coil a magnetomotive force which forces the flux around the magnetic circuit.

A true indication of the strength of the magnetomotive force of a coil can be obtained by multiplying the coil current in amperes by the number of coil turns, giving the number of *ampere-turns*. The greater the magnetomotive force (ampere-turns), the more flux it sends through the magnetic circuit, other things being equal. This corresponds to the Ohm's Law relationship that increasing the voltage will increase the current.

A clear understanding of ampereturns can prevent plenty of headaches if you ever wind experimental coils for electromagnets or relays. For example, adding more turns of the same size wire doesn't do any good if you still use the same coil voltage, because it doesn't change the ampere-turns. Here is the explanation: The extra wire increases the coil resistance, and this reduces the current just enough to offset the increase in turns. \vee

EXAMPLE: A 50-turn coil carrying 4 amperes gives 200 ampere-turns. Adding 50 more turns of the same wire gives 100 turns, but doubles the coil resistance. Therefore, if the coil voltage is the same as before, the extra turns just about cut the current in half, to 2 amperes. Multiplying 2 by 100 gives 200 ampere-turns, so the strength of the magnetic field stays the same even though the turns were doubled.

It should be pointed out that many radio coils are used in circuits where the resistance of the coil is negligibly small in comparison to the total resistance of the circuit. Here the coil current stays essentially the same even though we double the coil resistance, so increasing the number of turns increases the ampere-turns and gives a more powerful coil.

Remember, then, that the magnetomotive force (the force which sends magnetic flux around the magnetic circuit of a coil) depends upon the number of ampere-turns in the coil. The more ampere-turns we have, the more flux we get.

2. Reluctance. Just as resistance limits the amount of current flow in an



FIG. 4. Similarity between magnetic and electric circuits. The total reluctance is here the reluctance of the air through which the magnetic flux loops flow. Lines of force go out of the ends of the coil in all directions, but only a few of the lines can be shown in a diagram of this nature.

electric circuit, so does the flux path of every magnetic circuit have *reluctance* which offers opposition to magnetic flux. This reluctance is distributed along the entire path taken by the flux, but for study purposes it can be considered as concentrated at one point like a resistor. The comparison between reluctance and resistance is shown in Fig. 4. Here the magnetic path is through air. The lower we make the total reluctance of a magnetic circuit, the more flux we will get from a given magnetomotive force. This corresponds to the Ohm's Law relationship that lowering the resistance increases the current. On the other hand, increasing the reluctance cuts down the amount of flux.

From the standpoint of reluctance, all materials can be divided into two groups: 1. Non-magnetic materials; 2. Magnetic materials.

Non-Magnetic Materials. In this group, we have air and other non-magnetic materials, like paper, glass, bakelite, wood, brass, aluminum and copper. A coil having only non-magnetic materials in its core and in the form which supports the wire is an air-core coil, because the magnetic circuit has the high reluctance of air.

Magnetic Materials. This group includes iron, steel, ferrous (iron) alloys, and a few other metals and alloys which give magnetic circuits a lower reluctance than air.

Magnetic materials make it possible to secure large amounts of flux with only moderate values of magnetomotive force. A coil having magnetic material in its magnetic circuit is known as an iron-core coil.

How Reluctance Affects Flux. You can think about reluctance much as you do about resistance. Thus, the longer the magnetic circuit, the higher is its reluctance. (Compare with an electric circuit, where increasing the length of a wire increases its total resistance.) Conversely, shortening the path for flux decreases the total length of the magnetic circuit and thereby decreases the total reluctance.

The greater the cross-sectional area of a magnetic circuit, the lower is its reluctance. (In electricity, the thicker a wire, the lower is its resistance for a given length.) Conversely, reducing the cross-sectional area of the magnetic path increases the reluctance. The kind of magnetic material used has considerable bearing on the total circuit reluctance. Some materials are better magnetic conductors than others; the technician says the material has better *permeability*, just as copper has better conductivity than iron and therefore provides a lowerresistance path for current.

Permeability. The radio term which most conveniently expresses how much more flux we will obtain with a given magnetic material than with air is



Courtesy Continental Radio & Television Corp.

Coils are wound in many different shapes and sizes. In this cut-away view of the rear of a table-model Admiral receiver you see a basket weave coil known by the trade name "Aeroscope," serving as a built-in loop antenna.

permeability. In a practical sense, permeability expresses how well a material can conduct magnetic flux. (Permeability comes from permeate, which means to pass through.)

The permeability of a core material determines what the total reluctance of the core will be; when the permeability goes up, the reluctance goes down, and vice versa.

The permeability of air and all other non-magnetic materials is by agreement given the numerical value of 1. Magnetic materials all have higher permeability values than 1, with ac-
tual values ranging from about 50 all the way up to 10,000, or even higher for certain special alloys.

Air Gaps. In electrical circuits, we often find paths with different resistances in series. In magnetic circuits, we often-find different magnetic materials in series, as shown in Fig. 5. Here we have a magnetic circuit made chiefly of iron having low reluctance, but with a high-reluctance air gap at one point.

This is an actual case where most of the reluctance is concentrated at one point (in the air gap), with the rest of the circuit having such low reluc-



FIG. 5. The flux produced by this coil travels along a magnetic circuit consisting of an iron core and a short air gap. The total reluctance is but slightly higher than the air gap reluctance.

tance that it is comparable to the conducting wires in the electric circuit of Fig. 4.

3. Magnetic Flux. Finally, corresponding to current in an_electric cirouit we have magnetic flux (magnetic lines of force) in a magnetic circuit. Just as current is equal to voltage divided by resistance, so is flux equal to magnetomotive force divided by reluctance.

Summary. The Ohm's Law relationships for these three magnetic characteristics can be summarized as follows:

You can INCREASE the amount of flux either by increasing the magnetomotive force or decreasing the reluctance.

You can DECREASE the amount of flux either by decreasing the magnetomotive force or increasing the reluctance.

Every change in flux is thus due to a change either in magnetomotive force or reluctance. This thought is well worth keeping in mind during the remainder of your study of coils.

Using Coils to Produce Motion

There are a great many radio devices in which we use flux to produce motion. Typical examples are meters, loudspeakers and automatic switches (called relays).

Three Explanations. There are three fundamental ways of explaining how current-carrying coils can produce motion. Once you clearly understand these fundamental explanations, you can choose whichever is the most convenient for explaining the action of any motion-producing coil you may encounter.

1. Attraction and Repulsion of Magnetic Poles. The basic law of magnetism which you took up in an earlier lesson (like poles repel, and unlike poles attract) is highly convenient for explaining the operation of many types of coil devices.

Whenever a magnetic circuit made of iron or other magnetic material is broken by an air gap, the ends of the iron core on each side of the air gap will have opposite magnetic polarity due to the flux which flows through the iron magnetic circuit. The end which magnetic lines of force enter will be the S pole, and the end which lines of force leave will be the N pole. In other words, flux mag-





FIG. 6. Movement due to the attraction of unlike magnetic poles.

netizes any magnetic material through which it passes.

When you place a nail or other piece of iron on the pole of a permanent magnet, this nail acts like a magnet, and is capable of picking up small iron objects. The nail alone would not do this, so we say that the nail is *inductively magnetized* by the flux from the permanent magnet.

Since the core ends which are on opposite sides of an air gap have opposite polarity, they tend to attract each other in accordance with the law of magnetism. If a portion of the iron magnetic circuit of a coil is movable, as is the case in Fig. θ , this attraction across the air gap will result in a motion which tends to bring the Nand S poles closer together.

If an electromagnet is pivoted in the air gap between N and S poles in the manner shown in Fig. 7A, the repelling action between like poles will cause rotation in a clockwise direction as indicated by the arrows. If there



FIG. 7. Rotation due to repulsion of like poles (A) and attraction of unlike poles (B).

is no retarding force, this coil will finally come to rest at the position shown in Fig. 7B, in which the unlike poles are as close as they can get to each other. Magnetic poles thus attract and repel in the same manner regardless of whether they are on permanent magnets, electromagnets or inductively magnetized pieces of iron.

2. Position of Maximum Flux. In some cases it is more convenient to explain the action of a coil device by means of the following basic rule: Whenever a movable magnetic object is in or near a magnetic circuit, this object will tend to move to the position which gives maximum flux in the magnetic circuit.



FIG. 8. Pivoted objects in magnetic circuits take the positions shown here.

In the case of a movable iron object, this rule means that it will move to a position wherein its longest dimension is parallel to the magnetic lines of force in its vicinity. The movable object is then placing as much as possible of its length in the magnetic circuit, thereby lowering the reluctance of the magnetic circuit and giving maximum flux.

In the case of a movable electromagnet, this coil will take a position wherein its flux acts in the same direction as the original flux. The two fluxes then add, so as to give maximum flux as specified by the rule.

Thus, an iron object pivoted in the air gap between N and S poles of a

magnetic circuit will take the position shown in Fig. 8A, with its longest dimension parallel to the lines of force in the air gap. You will shortly learn that in this position the magnetic circuit also has minimum reluctance.

A current-carrying coil freely pivoted in the air gap between the N and S poles of a magnetic circuit will take the position shown in Fig. 8B. Its flux is then aiding the air-gap flux, thereby giving maximum flux.

3. Position of Minimum Reluctance. Last to consider, but equally as useful, is the rule which says: A movable magnetic object located in or near a magnetic circuit will tend to take a position which makes the reluctance of the magnetic circuit a minimum.

Minimum reluctance occurs when the greatest possible portion of the magnetic circuit is through iron or some other low-reluctance magnetic material. The movable piece of iron in Fig. $\boldsymbol{\theta}$ is moving to the position of minimum reluctance, because the reluctance is a minimum here when the air gap is closed up. Likewise, the pivoted iron object in Fig. $\boldsymbol{8A}$ is in the position of minimum reluctance, because it is in the position which gives the shortest possible air gaps.

Since lowering of reluctance makes flux increase, the position of minimum reluctance is identical with the position of maximum flux. It is simply more convenient sometimes to think of reluctance rather than flux. Likewise, since opposite poles at an air gap attract each other and tend to close up the air gap, thereby lowering the reluctance and increasing the flux, the law of magnetic attraction expresses exactly the same result as the other two ways of explaining the motion produced by a current-carrying coil.

Using Coils to Produce Voltage

Some microphones and some phonograph pick-ups are coil devices which use changes in magnetic lines of force (flux) to produce a voltage. The basic rule is: Whenever the flux linkages of a coil are changed, a voltage is induced in the coil. Before going into the explanation of this rule, let us first see what flux linkages are.

FLUX LINKAGES

Let us imagine that we have four lines of flux (4 magnetic lines of force) passing through a 5-turn coil, as illustrated in Fig. 9A. These lines of force can be produced by another coil or by a permanent magnet. (We will let straight lines represent flux, although we know that each of these straight lines is actually a portion of a complete loop of flux.) The electrical characteristics of this coil depend both on the number of turns in the coil and on the number of lines of flux passing through the coil. For convenience, radio men use the term *flux linkage* to describe this combination of flux and turns in a coil.



FIG. 9. Multiplying turns by lines gives flux linkages for a coil.



Courtesy Bud

There are plenty of coils in this interesting view of the r.f. section of a short-wave transmitter. The transmitter coils, without cores and with plastic spacer bars instead of forms, are prominent on top of the assembly. The box with a coil projecting from its back is a wavemeter, used to determine the approximate frequency at which the transmitter is operating. The bulb on top of the box lights up when the transmitter is properly adjusted.

Definition of Flux Linkage. One flux linkage is equal to one line of flux passing through (linking with) one turn of the coil. Thus, 1 line of flux passing through 10 turns gives 10 flux_ Inkages.

When all the flux passes through all of the turns of a coil, the total flux linkage is obtained by multiplying the number of turns in the coil by the number of magnetic lines of flux passing through all of the turns of the coil.

As an example, with our 4 lines passing through the 5 turns, we would have 4×5 , or 20 flux linkages. Clearly, it is much easier to say "20 flux linkages" than to say "4 lines of flux passing through 5 turns."

We can get the same number of flux linkages in a number of different ways, all of which will give exactly the same result. Thus, we can get 20 flux linkages by having 10 lines of force passing through 2 turns as in Fig. 9B, because 2×10 is 20. We can have 5 lines of force passing through 4 turns, because 5×4 is also 20. The important thing for you to remember is that flux (lines) times turns is equal to flux linkages.

Now Changing Flux Linkages. imagine that we have the 5 turns shown in Fig. 9A, but we increase the number of lines of flux from 4 to 8. This increases the flux linkages from 20 to 40, because 5×8 is 40. Whenever we increase the flux linkages of a. coil in this manner, a voltage is induced in the coil, and this voltage is known as an induced voltage.

If we decrease the number of flux linkages in a coil, such as by changing them from 40 back to 20 or from 40 all the way down to 0, we will again have a voltage induced in the coil. The strength of the induced voltage

depends not only upon how much we change the flux linkages, but also upon how long a time we take to make the change. If we change from 20 to 40 flux linkages in 1 second, we will get a certain voltage (the exact_value of this voltage is not important to us in this explanation). If, however, we change from 20 to 40 flux linkages in only 1/1000 of a second, we will get exactly 1000 times as much voltage as before. The faster the rate of change in flux linkages, the greater will be the induced voltage.

There is no need to figure exact values of flux linkage in practical radio work, because we are concerned chiefly with the effects produced when flux linkages *change* from instant to instant. The one thing you *should* remember, however, is this simple induced voltage rule:

Whenever the flux linkages of a coil are *changing*, a voltage is induced in the coil.

Lenz's Law for Coils. The induced voltage in a coil always acts in a definite direction. This direction at any given instant depends on just two things—on the direction of the original flux, and on whether its flux linkages are increasing or decreasing.

The exact relationship between these things is expressed by a famous electrical law. It is known as Lenz's Law, because he was the first to realize that the direction in which an induced voltage acts can always be predicted beforehand. The law in its basic form is expressed as follows:

LENZ'S LAW FOR COILS

The induced voltage always acts in such a direction that it tends to oppose the original change in flux linkages.



FIG, 10. Diagram illustrating how a conductor cutting across a magnetic field changes the flux linkages and gives an induced voltage.

If the flux linkages are *increasing*, the induced voltage will tend to prevent the flux from increasing. This means that the induced voltage will be in such a direction that it will tend to send through the coil a current which produces a magnetic flux which *opposes* the original coil flux. The induced voltage thus tends to prevent the increase in flux linkages.

On the other hand, if the flux linkages are *decreasing*, the induced voltage will be in such a direction that its current (when the coil circuit is complete) will produce a flux which *aids* the original flux, and thus tends to prevent the flux from *decreasing*.

METHODS OF PRODUCING CHANGES IN FLUX LINKAGES

The three basic methods of producing changes in the flux linkages of a coil are: 1. Cutting lines of force; 2. Changing the reluctance; 3. Changing the coil current. Each method will now be taken up in turn, after which actual radio devices using these methods will be studied.

1. Cutting Lines of Force. Imagine a steady magnetic field produced by a permanent magnet or electromagnet, with a conductor (wire) located in this field and connected to a voltmeter (V), as shown in Fig. 10. When we move this conductor upward through the flux in direction A, we induce a voltage into the conductor, and the voltmeter will indicate the presence of the voltage.

We get this induced voltage because motion of the conductor changes the flux linkages of the single-turn coil formed by the conductor and the voltmeter circuit. Moving the conductor upward, for example, *increases* the number of flux lines which are passing through the loop, thus increasing the flux linkages.

Sometimes it is easier to visualize and explain the production of an induced voltage by thinking of cutting lines of force. Whenever the conductor is moved through a line of force, it is cutting that line of force



The large heavy arrows indicate the direction of coil motion.

momentarily, and this produces an induced voltage in the conductor.

If we move the conductor parallel to the lines of force, as indicated by direction arrow B in Fig. 10, however, we do not get an induced voltage. We have neither cut lines of force nor changed the amount of flux passing through the coil, hence we have not changed the flux linkages of the coil.

Exactly the same explanation applies to Fig. 11, where an entire coil is being moved parallel to lines of flux. There is no cutting of lines, no change in flux linkages, and hence no induced voltage.

When we move a coil across a magnetic field as in Fig. 12, however, we are definitely cutting lines of force and changing the flux linkages of the coil, so we get an induced voltage.

Likewise, we cut lines of force and get an induced voltage when we rotate a coil in a magnetic field, as illustrated in Fig. 13. Here we are changing the amount of flux which passes through the various turns of the coil.

The faster we cut across lines of force with a conductor or coil, the faster we will be *changing* the flux linkages and the higher will be the induced voltage.

2. Changing the Reluctance. Motion of a movable iron portion of the magnetic circuit of a coil will change the reluctance. Any change in the reluctance of the magnetic circuit will change the amount of flux which passes through the coil, thus changing the flux linkages of the coil and inducing a voltage. Remember, whenever there is a change in flux linkage, a voltage will be induced.

3. Changing the Coil Current. When two coils are arranged as shown in Fig. 14, whereby the flux produced by current-carrying coil L passes through coil L_1 , we can induce a voltage in coil L_1 simply by changing the current through coil L. By changing



FIG. 14. This diagram shows how a change in current in one coil can induce a voltage in another coil.

the current, we change the amount of flux produced by L. This changes the amount of flux passing through L_i , thereby changing its flux linkages and producing the induced voltage. The faster the current is changed, the faster the flux linkages will change and the greater will be the induced voltage. Changes in current are usually secured in radio coil devices simply by sending alternating current through the flux-producing coil. An alternating current is varying continually in value from zero to maximum values in either direction, so it makes flux linkages change continually.

The higher the frequency, the more cycles of change there are per second and the faster is the current changing at all times. Therefore, increasing the frequency makes current, flux, and flux linkages all change faster, giving a higher induced voltage.

EXAMPLES

The best way to understand (not memorize) basic radio principles is by seeing how they are app'led in actual radio parts. The following examples of voltage-producing devices will therefore familiarize you with these important principles and at the same time make you acquainted with the



FIG. 15. Simplified cross-section views of a dynamic microphone.

construction and operation of some highly important radio parts.

Dynamic Microphone. This is an example of a voltage-producing radio device in which the changes in flux linkages are produced by making a coil cut magnetic lines of force.

The general construction of a dynamic microphone is shown in Fig. 15A. A strong fixed magnetic field is produced by a permanent magnet inserted in the central soft iron core. Two return paths through iron, with an air gap in each one, are provided to complete the magnetic circuit of the permanent magnet from its lower end to its upper end.



Modern Dynamic Microphones

The voice coil is attached to a flexible metal diaphragm in such a way that this coil moves in and out of the flux in the air gap whenever the diaphragm is moved up and down by sound waves.

When the voice coil is at the normal at-rest position determined by the springiness of the diaphragm (Fig. 15A), only a few turns of the voice coil are in the air gap. Only a part of the total air gap flux passes through these few turns, hence the number of flux linkages is quite small. These flux linkages are constant when the coil is at rest, hence no voltage is induced in the coil.

When sound waves push the diaphragm down, the coil moves farther into the air gap, as in Fig. 15B, cutting magnetic lines of force and thus *increasing* the flux linkages of the voice coil. This induces a voltage in the voice coil.

When sound waves pull the diaphragm up, the coil moves out of the air gap as in Fig. 15C, cutting those lines of force which formerly were passing through some of the coil turns. Now we are *reducing* the flux linkages, and again we have an induced voltage. The voice coil thus cuts magnetic lines of force and changes its flux linkages whenever the coil is in motion. As a result, the voltage induced in the coil is an a.f. voltage corresponding to the sound wave which causes the movements of the diaphragm and voice coil.

Magnetic Phono Pick-up. Here is an example of a unit which depends for its action upon a *change in reluctance*. The series of diagrams in Fig. 16 will help you to understand the construction and operation of one of these units.

This phono pick-up unit is designed for lateral-cut phonograph records, where the groove is always the same depth but wiggles from side to side in accordance with the audio signal. In following the wavy path of a groove, the phonograph needle moves from side to side. This causes the pivoted iron armature to rock back and forth between the two U-shaped soft iron pieces, one of which is on the inside of each pole of the horseshoe-shaped permanent magnet.

When the groove in the phonograph record is straight, as during a moment of silence in a recording, the pivoted armature has the mid-position shown in Fig. 16A. Flux now divides about equally over the two paths from the N poles to the S poles because both paths have about the same reluctance, and there is essentially no flux traveling vertically through the armature. As a result, there is no flux passing through the coil and no flux linkages in the coil.

When the needle tilts the armature to one side, as shown in Fig. 16B, the lowest-reluctance path is that shown by the dotted arrows in this diagram. The farther the top of the armature tilts to the left, the lower becomes the reluctance and the more flux flows through the armature. This changing flux induces a voltage in the coil which surrounds the armature. The flux is changing continually because the phonograph needle is moving continually from side to side as it rides in the groove.

When the needle tilts the armature the other way, the flux will travel in the opposite direction through the armature, as indicated in Fig. 16C.



FIG. 16. Cross-section views illustrating the basic operating principles of a magnetic phono pick-up.

When the needle moves continually back and forth between the positions of Figs. 16B and 16C during the playing of a record, the continually varying and reversing flux in the armature and in the coil keeps the flux linkages of the coil changing continually, thereby inducing in the coil an a.f. voltage corresponding to the sounds recorded on the phonograph record.

Rubber dampers prevent the top of the armature from "sticking" in any





A.F. TRANSFORMERS POWER TRANSFORMERS Examples of iron-core transformers you will encounter in radio receivers.

one position, and also prevent it from vibrating too greatly under certain conditions.

Transformer. The best-known example of a radio part in which changes in flux linkages are secured by changing the coil current is the *transformer*. The basic construction of one type of transformer, an iron-core unit, is shown in Fig. 17 to illustrate the basic principles involved.

Two coils are wound around the central part of the iron core. These coils could be placed side by side as shown in the diagram. In actual units, however, one is usually wound over the other, with insulating cloth between, because this gives lower manufacturing costs without affecting performance.

The winding through which we send the current from our voltage source is called the *primary winding*, and is usually marked PRI. or P. The other winding, in which the voltage is induced, is called the *secondary winding* and is usually marked SEC. or S. A transformer will have only one primary winding, but can have any number of secondary windings.

When steady direct current is sent through the primary winding, this current produces magnetic lines of force which take the paths shown in Fig. 17. This flux passes through the secondary winding and gives it a definite number of flux linkages. With direct current, however, the number of flux linkages does not change, and hence no voltage is induced in the secondary winding.

Whenever we change the value of the primary current, the flux in the iron core changes accordingly, and hence the flux linkages of the secondary winding change. As a result, a voltage is induced in the secondary whenever the primary current changes.

When an alternating current is sent through the primary, we have a continually varying primary current and correspondingly varying flux linkages in the secondary winding. The voltage induced in the secondary under this condition is an a.c. voltage of the same frequency. A transformer thus transfers an a.c. voltage from one circuit to another without wire connections.





Basic Idea of Inductance

Self-Induced Voltage. Up to the present time, we have studied what happened when coils were placed in magnetic fields produced by other devices. However, a coil can be entirely by itself, far away from other magnetic fields, and still have an induced voltage. Here is how this "selfinduced" voltage is produced.

. 1

When current is sent through a single coil, the coil produces its own magnetic field. And when this coil current changes, due to a change in the applied voltage, the magnetic field also changes in strength. Naturally, this magnetic field links with the coil, so we have changing flux linkages in the coil. This means that there is an induced voltage in the coil, produced by the coil itself and hence called a *self-induced voltage*. Lenz's Law says that this *self-induced voltage* must be in such a direction that it *opposes* the original change.

As an example, when we increase the coil current by boosting the applied voltage, we get an induced voltage which opposes the applied voltage and thus opposes the increase in applied voltage. The important thing for you to realize is that the induced voltage always opposes the change which is producing the induced voltage. If we didn't have this natural opposition to changes—if instead the induced voltage aided the change—then the induced voltage would build up to tremendously high values and destroy the coil.

Inductance. The characteristic which determines how much voltage will be induced in a coil by a given change in current is called inductance. Inductance is one factor which determines how much the self-induced voltage will be. More specifically. the electrical size or inductance of a coil determines how much *change* in flux linkages will be obtained with a given change in coil current. The higher the inductance, the greater will be the change in flux linkages for a given current change.

Units of Inductance. The basic unit of inductance is the *henry*. It is named after Professor Joseph Henry, an outstanding American scientist who announced the results of his coil experiments in 1832. The henry specifies the amount of flux linkages produced by an ampere of current. If one ampere produces 100,000,000 flux linkages in an air-core coil, the coil has an inductance of one henry.

All coils, including iron-core coils, are said to have an inductance of one henry when a current change of one



Courtesy General Electric Co.

Inside the cardboard housing of the "Beam-A-Scope" in this General Electric console receiver is a large coil which serves as a bufit-in loop antenna. The three cylindrical shield cans on top of the chassis contain r.f. and i.f. coils, and the two loudspeakers have iron-core coils. ampere in one second produces a selfinduced voltage of one volt.

Although some of the iron-core coils used in radio have inductance values ranging as high as 1000 henrys, aircore radio coils have inductance values of only a small fraction of a henry. For convenience in specifying inductance values of air-core coils and small iron-core coils, we frequently use another unit of inductance called the *millihenry*. One millihenry is equal to one-thousandth of a henry, hence there are 1000 millihenrys in one henry. Here are some examples: 500 millihenrys is equal to .5 henry; 9000 millihenrys is equal to 9 henrys.

A still smaller unit of inductance occasionally encountered in radio is the *microhenry*. It is equal to onemillionth of a henry. It takes one thousand microhenrys to make one millihenry.

Factors Affecting Inductance. The inductance of a coil is determined by the number of turns in the coil, by the shape of the coil, and by the material used in the core of the coil. It is possible to calculate by means of formulas the inductance which will be obtained with any combination of these factors, but practical radio men do not particularly need to know such design procedures. \checkmark

It is entirely sufficient to know that when you increase the number of turns in a coil, the inductance will increase likewise. If you reduce the number of turns, the inductance will go down. Furthermore, the change in inductance will be somewhat faster than the change in turns.

It should be pointed out that inductance is not necessarily limited to coils. Even a straight wire has inductance; in fact, straight wires or straight pipes are actually used as inductances in ultra-high-frequency equipment. The inductance value is too small for the type of radio receivers we use in our homes, but becomes effective and useful at ultra-high frequencies.

A. C. Voltage Drop of a Coil. We have learned that whenever the flux linkages of a coil are changed in any way, a voltage is induced in the coil. Therefore, if we send alternating current through a single coil, the continually changing flux linkages will cause an a.c. voltage to be induced in the coil itself.



Courtesy Barber-Colman Co.

A single coil wound on an iron core provides the varying magnetic field which operates this small a.c. motor, used to drive the magnetic tuning unit of a radio receiver. The "squirrelcage" armature at the upper right fits into the hole above the coil, being supported there by two bearing assemblies like that shown at the lower right.

Since an induced voltage always opposes whatever force is producing it, a self-induced a.c. voltage in a coil opposes the a.c. source voltage. This is why a self-induced voltage is sometimes known as a back e.m.f. or counter e.m.f.

The back e.m.f. in a coil is the a.c. voltage drop appearing across the coil. Just as the voltage drop across a resistor is due to current flowing through the opposition in ohms of a resistor, so is the a.c. voltage drop (back e.m.f.) of a coil due to the opposition in ohms which the coil offers to alternating current. This coil opposition is known



Courtesy Henry L. Crowley & Co., Inc.

These two molded cup-shaped outer pieces and the cylindrical iron core of pulverized iron will, when assembled, form a complete magnetic core for the r.f. choke coil, greatly increasing its inductance.

as *inductive reactance*, and is well worth thorough study now because it determines how much alternating current will flow through a coil in a particular radio circuit.

Inductive Reactance. The opposition in ohms which a coil offers to the flow of alternating current varies with frequency. This explains why the frequency.value is included in the following rule for figuring the inductive reactance of a coil:

The inductive reactance in ohms of a coil is equal to the inductance of the coil in henrys multiplied by 6.28 times the frequency in cycles.

By using letter notations for some of these terms, we can express this relationship more conveniently by an equation or formula. The notation X_L is used to represent <u>inductive re-</u> actance. The smaller letter f is used to represent frequency. The letter Lis used in formulas to represent inductance. With these notations, the inductive reactance in ohms of a coil can be expressed by means of the following formula:

$X_{\rm L} = 6.28 \times f \times L$

You will occasionally find this formula written as $X_L = 2\pi f L$; π is the

Greek letter pi (pronounced *pie*) which is commonly used to represent the mathematical number 3.14, and 2π is 2×3.14 , or 6.28, the number in our first formula.

Example: Suppose we wanted to know the inductive reactance of a 30-henry choke coil at 120 cycles. The formula is:

$$X_{\rm L} = 6.28 \times f \times L$$

Substituting 120 for f and 30 for L gives

$$X_{\rm L} = 6.28 \times 120 \times 30$$

Multiplying these three numbers together gives 22,608 as the value of X_L . This coil would therefore have an opposition (inductive reactance) of about 22,600 ohms in a 120-cycle alternating current circuit.



Courtesy Henry L. Crowley & Co., Inc.

Examples of pulverized iron cores used in adjustable r.f. coils. Each screw is slotted at the end for adjusting purposes; rotating the screw with a screwdriver moves the core into or out of the coil, thereby changing the inductance. This method of changing the inductance of a coil is often called permeability tuning.

At 60 cycles, the inductive reactance of this coil would be $6.28 \times 60 \times 30$, or 11,300 ohms, which is exactly half the reactance at 120 cycles. At 120,-000 cycles, the inductive reactance would be $6.28 \times 120,000 \times 30$, or 22,600,000 ohms. Inductive reactance thus varies directly with frequency. These examples are given not because you will solve such problems as a technician, but to show you how reactance varies with frequency, and to show that the reactance also varies with the inductance of a coil.

Just as a matter of comparison, we might point out that in a direct current circuit, the only opposition this 30-henry coil would offer to the flow of current would be that due to the resistance of the copper wire in the coil-about 400 ohms for a coil of this This d.c. resistance will also size. affect alternating current, and will produce an a.c. voltage drop. This resistance value is negligibly small in comparison to the 22,600-ohm inductive reactance at 120 cycles, so we normally neglect the a.c. voltage drop due to the resistance of the coil.

Ohm's Law for Coils. Ohm's Law tells us how much alternating current will flow through a particular coil under given conditions, just as it does for resistors in direct current circuits. We need to know only the inductive reactance of the coil in ohms and the effective value of the a.c. voltage applied to the coil. The effective alternating current value in amperes is then equal to the a.c. voltage in volts divided by the reactance in ohms. (This assumes that the d.c. resistance of the coil is negligibly small in comparison to its reactance.)

The simple formula is $I = E \div X_L$; you will recognize this as being very similar to $I = E \div R$, the d.c. version of this Ohm's Law formula.

Example: A coil having a reactance of 6000 ohms is to be connected across a 120-volt a.c. power line. What current will flow through the coil?

Since I is equal to $E \div X_{L}$, we divide 120 by 6000, and get .02 ampere as the a.c. coil current.

The other two Ohm's Law formulas for coils having negligibly small resistance are $E = I \times X_L$ (for use when you want to figure the a.c. voltage drop across a coil) and $X_L =$ $E \div I$ (for use when you want to figure the inductive reactance of a coil).

Mutually Coupled Coils. When the two coils of a transformer are wound side by side or one over the other on a paper or fiber form, without any magnetic material in the core, we have an *air-core transformer*, and the two coils are mutually coupled through their magnetic fields. Aircore units are usually called *r.f. transformers* or *i.f. transformers*, because they are used chiefly in radio fre-



quency (r. f.) and intermediate frequency (i.f.) circuits.

Radio men often speak of r.f. coils when they mean r.f. transformers. An r.f. transformer does consist of two or more r.f. coils, so there is some justification for this. Anyway, the terms are a part of the radio man's everyday language, so we might as well accept them.

In two coils which are mutually coupled, the voltage induced in the second coil depends on three things: 1. The *frequency* of the primary current; 2. The effective value of the primary current; 3. The mutual inductance of the two coils. Increasing any one of these three things will increase the induced voltage in the secondary, as explained later.

Effect of Frequency. The higher the frequency of the primary current, the faster will the flux be changing and the greater will be the induced voltage in the secondary.

Effect of Current. The higher the value of primary current, the more flux it will produce. Since this flux varies from a maximum value in one direction to zero and then to a maximum in the other direction, more flux means a greater change in flux per unit of time and a higher induced voltage in the secondary.

Mutual Inductance. This is a factor which expresses in a practical manner the amount of coupling (flux linkage) between the primary and secondary coils of an air-core transformer. The greater the mutual inductance value, the greater will be the voltage induced in the secondary of the transformer by a given change in primary current.

Mutual inductance depends upon the sizes of both air-core coils, the number of turns on each coil, their relative positions and their distances apart.

Mutual inductance is specified in henrys just like ordinary inductance, and is represented in formulas by the letter M. The higher the value of mutual inductance, the greater will be the induced voltage in the secondary coil.

Sometimes you will find mutual inductance defined as follows: When a change of one ampere per second in primary current produces one volt in the secondary, the mutual inductance is one henry. This applies to any mutually-coupled coil.



FIG. 18. Coils combine exactly like resistors

COILS IN SERIES AND PARALLEL

It is worth knowing in a general manner what happens when coils are combined in series or in parallel, the coils being far enough apart so that their magnetic fields do not affect each other, or shields being used to prevent interaction of the coils.

When coils are connected together in series, as shown in Fig. 18A, the combined inductance is the *sum* of the individual inductances.

When coils are connected together in parallel, as shown in Fig. 18B, the combined inductance is *less than* the smallest inductance in the group.

These rules for combinations of coils are exactly the same as for combina-



Air-core transformers for short-wave transmitters. The amount of coupling between the two coils is varied in the left-hand unit by rotating a small inner coil, and in the right-hand unit by moving a link (one or two-turn coil) between the two coils.

tions of resistors. This fact makes it easy for you to remember how coils combine.

Mutually-Coupled Coils in Series. When two coils are connected in series and are close enough together so that mutual inductance exists, we have interaction between the coils, and the combined inductance will be either increased or decreased by a factor equal to twice the value of the mutual inductance in henrys. Thus, if coil connections are such that the magnetic fields of the two coils aid each other. the combined inductance will be equal to $L_1 + L_2 + 2M$. If the connections are such that the magnetic fields of the two coils oppose each other, the combined inductance will be $L_1 + L_2$



FIG. 19. Leakage flux. In the average coil, the leakage flux is only a small proportion of the total useful flux.

-2M. Reversing the connections to one of the coils will reverse the effect of the mutual inductance factor.

Leakage Flux. Flux which is produced by the primary winding of a transformer but does not produce the maximum possible number of flux linkages with the secondary winding is called *leakage flux*. Two mutuallycoupled air-core coils do not give the maximum possible transfer of voltage from primary to secondary because of this leakage flux.

In a single coil by itself, the flux which is produced by some of the turns but does not link with all of the turns of the coil is called *leakage flux*. This is basically the reason why certain shapes and sizes of coils give greater inductance for a given length of wire.

The_name leakage-flux is used because this flux "leaks" back around a path whereby it does not do its full quota of useful work. Examples of both types of flux are shown in Fig. 19. You will frequently encounter the subject of leakage flux as you continue your study of radio.

VARIABLE INDUCTANCES

In some receiver and transmitter circuits, it is necessary to provide means for changing the inductance value of the coil in a radio circuit. Thus, in receivers having more than one band, different inductance values are needed for each band. This can be done in a number of different ways, which we will now consider, one at a time.

Changing Entire Coils. One way to change quickly from one inductance value to another is by means of plugin coils, which fit into sockets in the chassis and can be changed just like tubes are changed. Plug-in coils are used today chiefly in experimental receivers, communications receivers and transmitters, for band-changing purposes so as to permit reception or transmitter operation in a particular band of frequencies.

A more convenient way of changing, coils, used in many modern all-wave



Examples of plug-in coils used in experimental short-wave receivers.

receivers and communication transmitters, is by means of a band-changing switch which disconnects one set of coils completely and connects another set in its place when the switch knob is rotated. Sometimes the coils are provided with taps, and the bandchanging switch makes connections to the taps which place the desired numbers of turns in the circuits.

Variometers. The inductance of a coil can be varied gradually between two values with a variometer. This is simply a variable inductance in which one coil is rotated inside a larger coil. The two coils are connected in series. Rotation of the inside coil changes the amount of coupling between the two coils, changing the mutual inductance of the two coils and thus changing their combined in-



Transmitter band-switching arrangement which makes it possible to connect any one of three coils into the circuit merely by turning the switch. Many all-wave receivers have similar band-switching arrangements, using smaller coils and more compact switches.

ductance. Variometers were once widely used in receivers for tuning purposes, and are still to be found in some types of radio transmitters.

Permeability Tuning. The inductance of a coil can be varied by moving a small cylindrical core of pulverized iron into or out of the coil. This varies the reluctance of the magnetic circuit of the coil, which changes the amount of flux the coil can produce, and thus varies the inductance.

This method of varying inductance



R.F. transformer with adjustable pulverized iron core. Tightening the adjusting screw raises the core out of the primary coil, thereby reducing the amount of coupling between the two coils.

is usually known as *permeability tuning*, because as we include more of the iron in the magnetic path of the coil, we change the permeability of the magnetic path. Some modern receivers are now using variable inductances like this with fixed condensers for tuning purposes, in place of variable condensers with fixed inductances.

RESISTANCE OF A COIL

When a coil is connected to a d.c. voltage source, direct current flows through the coil. The value of this steady direct current depends only on the *d.c. resistance* of the coil, which is simply the resistance of the copper wire used in constructing the coil. (The inductive reactance of the coil has no effect whatsoever on the amount of direct current which flows through the coil, as reactance exists and offers opposition only in a.c. circuits.)

Although copper wire has a lower resistance per unit length and thickness than almost any other material you encounter in radio, coils are sometimes made with so much fine copper wire that they have d.c. resistance values even higher than 1000 ohms. On the other hand, some coils are made from only a few turns of large wire, so their d.c. resistance is considerably less than 1 ohm.

A knowledge of d.c. resistance values of coils is extremely useful when hunting for trouble in radio equipment. Just by comparing the measured d.c. resistance of a coil with the value specified on the circuit diagram, you



FIG. 20. The d.c. resistance values of coils are often given on circuit diagrams of radio receivers, much in the manner shown here.

may be able to tell whether or not a coil is defective.

You will often find resistance values specified near the coil symbols on circuit diagrams, much in the manner shown in Fig. 20 for typical coils of different types. This information is ' particularly useful when checking resistance or continuity in a circuit with an ohmmeter in order to locate a defective part, as you will learn when you begin mastering radio servicing techniques.

Another practical use for d.c. resistance values of coils is in identifying the coils in the various bands of an all-wave receiver. Coils for shortwave bands always_have less induct= ance than broadcast band coils. This means that short-wave coils have fewer turns, less wire and less resistance. The tuning coil having the lowest resistance can therefore be identified as belonging to the lowest shortwave band (the highest-frequency band). The tuning coil having the most resistance will belong to the broadcast band.

Using Ohm's Law. A coil in a direct current circuit acts exactly like a resistance insofar as the d.c. voltage source is concerned. This means you can use Ohm's Law to figure voltage, current and resistance values for coils in d.c. circuits, just as you do for resistors. For example, to figure the amount of direct current which will flow through a coil, you divide the d.c. voltage applied to the coil by the d.c. resistance of the coil $(I = E \div R)$.

Coils Can Get Hot. In the previous lesson, you learned that whenever current flows through a resistance, heat is produced. The copper wire in a coil has resistance, and therefore heats up exactly like a resistor. If you overload a coil by sending too much direct current through it, the wire in the coil melts and burns out just like an overloaded wire-wound resistor. When coils are sealed with pitch, as is the case in some output transformers and filter chokes, overloading may melt the sealing compound, causing smoke and a characteristic burned odor.

Magnet Wire. Coils are usually wound with one or more layers of insulated copper wire. You will find three types of insulation in common use—enamel, cotton and silk, used singly or in combinations like silk



Examples of choke coils you will encounter in radio receivers and transmitters. Single coils by themselves are called choke coils by radio men, and are used individually in radio circuits to offer opposition to alternating current without appreciably hindering the flow of direct current. Basket-weave windings (like that of the unshielded r.f. choke) and criss-cross windings give low distributed capacity (explained later), thereby improving the effectiveness of the coil at high radio frequencies.

over enamel and cotton over enamel. Wire in the sizes and types used for coils is popularly known as magnet wire, because at one time such wire was used chiefly for electromagnets.

Coils are sometimes baked in a hot oven during manufacture to drive off moisture, then dipped in insulating varnish or molten wax to make them moisture-proof and to hold the turns of wire in position despite rough handling.

NON-INDUCTIVE RESISTORS

Wire-wound resistors made by winding nichrome resistance wire around an insulating form are really small coils having high resistance. Oftentimes, the coil characteristics of these wire-wound resistors are highly undesirable. This is particularly true in the high frequency radio circuits used in frequency modulation receivers and transmitters, in television receivers and transmitters, in ultra-high frequency police radio equipment, and in radio airplane locators. For these circuits we must use non-inductive resistors, which have no coil characteristics.

Carbon and metallized resistors are considered as non-inductive resistors, for each one is a single straight rod without any turns or coils, having negligible inductance.

Special windings are used on wirewound resistors to get non-inductive characteristics. One example is the so-called "hairpin-type winding" shown in Fig. 21A. In this winding.



HAIRPIN-TYPE Non-Inductive Winding

 (\mathbf{A})



B SECTION-TYPE Non-Inductive Winding



the electron flow is in opposite directions through any two adjacent wires, hence the magnetic fields produced by the adjacent wires are in opposite directions and cancel each other. The result is zero inductance.

Another type of non-inductive winding is shown in Fig. 21B. Here the resistor is made of an even number of sections, connected so current flows in opposite directions through adjacent sections. The magnetic fields of adjacent sections cancel each other. and again we have zero inductance.

Simple Coil-Resistor Circuit

We have already seen that when a coil alone is connected to an a.c. voltage source, its reactance determines what current will flow. We now some to a practical case where a coil is used in series with a resistor in an a.c. series circuit.

If we connect a coil and a lamp in series to a 115-volt d.c. source, as in Fig. 22A, then connect d.c. voltmeters to measure the voltage drops across the coil and lamp, we will find that these voltages check with Kirchhoff's Voltage Law. In other words, the voltage drops will add up to 115 volts, the source voltage.

Suppose, however, that we change to a 115-volt a.c. source, as in Fig. 22B. With a.c. voltmeters being used now to measure the voltage drops, we find that the two voltage drops add up to 154 volts, which is much higher than the source voltage. Clearly, we cannot apply Kirchhoff's Voltage Law directly to a.c. circuits.

We know this law must hold true because it is a fundamental radio law, so let us now see why it doesn't apply directly. Let us also investigate the special procedure a radio design engineer might have to use in order to make the law apply. This problem brings us for the first time to the subject of phase.

Phase. The voltage drops across the coil and lamp in Fig. 22B do not add up to the a.c. source voltage because these a.c. voltage drops do not reach corresponding peak values at the same instant of time in each cycle. In other words, one a.c. voltage may be at the zero point in its cycle at the instant when the other a.c. voltage is at its maximum or peak value. We encounter this situation whenever we use coils or condensers along with resistors in radio circuits, and we say that there is a *phase difference* between these a.c. voltage drops.

The only time we can add directly the a.c. voltages which a.c. meters measure is when these voltages are in phase with each other (when all reach corresponding peak values in each cycle at the same instant of time, and all drop to zero together). This occurs only when all the parts in the circuit are identical (all are resistors, all are perfect coils or all are condensers). Let us see how phase enters into the picture when we use a coil in an a.c. circuit.

First of all, we can definitely say that the same value of alternating current flows through both the coil and the lamp in our a.c. circuit of



FIG. 22. Circuits illustrating the effects of phase in a.c. coil circuits.

Fig. 22B, because these two parts are in series. The curve in Fig. 23A shows how this current varies during each complete cycle. On diagrams of this type, time is always assumed to be zero at the extreme left of the curve, so the beginning of the cycle is at the left on the diagram. Time therefore increases to the right, and the end of the cycle is at the right.

A lamp is a resistance, so this current produces across the lamp an a.c. voltage drop which is *in phase* with the current. The resistor voltage curve is shown in Fig. 23B for the same interval of time covered by the current curve in Fig. 23A. Compare these two curves, and you will see that voltage and current values for a resistor reach corresponding peak values in each cycle at exactly the same instant of time.

The a.c. voltage across the coil is definitely not in phase with the circuit current. Technically speaking, the



FIG. 23. These two curves show that voltage and current values for a resistor reach corresponding values at exactly the same instant of time. This means that voltage and current for a resistor are IN PHASE with each other.

current in a perfect coil always lags the coil voltage by 90° (one-quarter cycle). The reason for this can be explained in a few sentences, with the aid of the circuit current curve in Fig. 24A. First of all, we must remember that the a.c. voltage drop of a coil depends upon how fast the coil current is changing.

Examining the current curve in Fig. 24A, we see that the current changes



FIG. 24. The basic fact that a coil voltage is OUT OF PHASE with the coil current is shown by these two curves.

very little-hardly at all-during time interval a. The current on this diagram is represented by the distance a point on the curve is from the horizontal reference line. Since points 1 and 2 are both almost the same distance above this line, the current is essentially the same at 2. the end of time interval a, as at the beginning (1). When the current doesn't change. the flux doesn't change, and hence the flux linkages do not change. No changes in flux linkages mean no induced voltage to act on an a.c. voltmeter connected across the coil, which explains why we find the coil voltage to be zero when we trace downward from point 2 to the coil voltage curve in Fig. 24B.

At time interval b (Fig. 24A), however, the current is changing very rapidly (it drops from the value at point 3 all the way down to zero at point 4).

The steeper (more vertical) the curve, the more change there is in current during a given interval of time. We therefore expect the a.c.



Courtesy Sonsitive Research Instrument Co

The number of magnetic lines of force passing through a given area can actually be measured with this special instrument known as a fluxmeter. When the exploring search coil (below the instrument) is rotated or moved in a magnetic field, the resulting change in the flux linkage of this coil produces an induced voltage which sends current through the meter. The meter scale is calibrated to indicate lines of force. (On this instrument, each small division on the scale represents a change of 10,000 lines of force in a single-turn exploring coil.)

coil voltage drop to be a maximum at the time of point 4, where the curve is most nearly vertical, and we find this to be true when we trace downward to the coil voltage curve.

Continuing in this same way for other instants of time, we will find that when coil current is represented by the curve in Fig. 24A, the coil voltage will vary as shown by the curve in Fig. 24B.

If we carefully compare the two curves in Fig. 24, we will find that the coil voltage curve reaches its peak value exactly one-quarter cycle before the current curve reaches a corresponding peak. For example, at zero eurrent the coil voltage is at its upper peak (5), but the current does not reach its upper peak (2) until a quarter-cycle later. This is why we say that the coil voltage leads the current by one-quarter cycle, which is 90°. (One cycle-is 360°, so one-quarter cycle is 90°, pronounced ninety degrees.) This holds true for all perfect coils.

Of course, we can express this relationship in another way if we prefer: The coil current lags 90° behind the coil voltage. Saying current lags voltage is exactly the same as saying voltage leads current.

Since the coil voltage is out of phase with the current, and the resistor voltage is in phase with the current, we arrive at the fact that the coil and resistor voltages are out of phase with each other. A.C. voltages as read by a meter can be added directly only when they are in phase with each other, and consequently we cannot add these two a.c. voltages directly.

Always remember that a.c. meters measure *effective* values of a current or voltage which is continually varying. The meters read the same even though the current through them is varying, because meter coils cannot respond to such fast variations in current.

If we had some means of measuring the exact values of our a.c. voltages all at the same instant, we would find that the instantaneous voltage drop values would add up to the exact value of the source voltage at that instant. We have no convenient meters for this, however, and there is no practical need for measurements of instantaneous a.c. values, so let us go on now and learn the correct way for combining outof-phase a.c. voltages.

Vectors. We have found that the coil and resistor (lamp) voltages reach their peaks at different times in each cycle. The simplest way to show the phase relationships of all voltages and currents in an a.c. circuit is by means of a single diagram in which rotating arrows represent the a.c. voltages and currents. These arrows are called *vectors*.

The length of each vector indicates the *amount* of voltage or current, and the position of the vector with respect to the starting or reference line indicates the phase of the voltage or current with respect to the other voltages and currents.

Each complete revolution of a vector represents one a.c. cycle. By general agreement among radio and electrical men, vectors are always assumed to be rotating *counter-clockwise*, op-



FIG. 25. Vector diagrams like these are easy ways of showing how continually varying currents and voltages in a.c. circuits are related to each other.

posite to the direction in which the hands of a clock move. Furthermore, the starting or reference position for all vectors is the vector position shown in Fig. 25A, which is a line going to the right horizontally from the center (0) of the vector diagram.

In a series circuit, we almost always use current for our reference vector. To make the vector in Fig. 25A represent the circuit current, we make the length of the vector proportional to the effective current value which would be indicated by an a.c. ammeter. We then put on the arrow head, and label it with the capital letter I to indicate that it represents current.

The voltage drop across the lamp is



FIG. 26. These simple vector diagrams tell the entire story about how the current and voltage are related to each other in the a.c. coil resistance circuit of Fig. 22B.

in phase with the current, so the voltage drop vector $V_{\mathbf{R}}$ will have exactly the same position as the current vector. We therefore draw voltage drop vector $V_{\rm R}$ along the reference line, right over the current vector and in a convenient length which is proportional to the effective a.c. value of the resistor voltage drop, just as in Fig. 25B. Thus, we might let one inch of length along the vector line represent 5 volts, 20 volts or any other value which gives a convenient size of vector diagram. This simple diagram now tells us at a glance that the voltage and current for the lamp are in phase with each other. Incidentally, a vector diagram with overlapping vector arrows like this would be obtained for any resistor in any a.c. circuit.

Next comes the coil voltage vector, which we know must be 90° ahead of the current. One complete vector revolution is one cycle or 360° , so 90° will be one-quarter of a revolution. We must go one-quarter cycle counterclockwise to get ahead 90° , so we draw coil voltage vector $V_{\rm L}$ straight up from 0, as in Fig. 26A. Coil voltage and current_vectors $_{90^{\circ}}$ apart like this would be obtained for any perfect coil in any a.c. circuit.

Only one voltage remains to be placed on our vector diagram—the source voltage. And now we learn the secret of making Kirchhoff's Voltage Law apply to our a.c. circuit: If we take the *vector sum* of the voltage drops, they will add up to the source voltage E.

To get the vector sum of voltage drops $V_{\rm L}$ and $V_{\rm R}$ on our diagram in Fig. 26B, we simply draw in dotted lines to complete a rectangle, then draw the diagonal of this rectangle from 0. This diagonal line is the vector sum of the coil and resistor voltage drops, and is also equal to the source voltage, so we place an arrow on its other end and label it E. Since this vector is ahead of current vector I (remember that we have assumed counter-clockwise rotation), we say that source voltage E leads the circuit current. If we prefer to express it the other way, we would say that the current I lags the source voltage.

If we made the lengths of the voltage vectors in Fig. 26B proportional to the actual effective a.c. voltage values (for instance, if we let each inch of length along the vector represent a definite amount of voltage), then measured the length of voltage vector E in Fig. 26B and converted it back to volts, we would find that it was exactly equal to the source voltage value. Kirchhoff's Voltage Law holds true for all a.c. circuits if we apply it in this way so as to take phase into account.

The vector diagram in Fig. 26B thus tells us the complete story about voltage and current relationships in our coil-resistor circuit shown in Figs. 22B and 26B.

Importance of Phase. A general knowledge of phase and vector diagrams will help you to understand the actions of coils and condensers in a.c. radio circuits, and will make you a better-than-average radio man. You will understand why you do certain things when making adjustments or repairs, instead of just blindly following instructions. It is the men who know why and how who command the highest salaries in the modern world of radio.

Later in your course, you will learn that the opposition of a coil can be balanced or cancelled by the opposition of a condenser because of phase; you will find that hum can be balanced out of a receiver because of phase; pictures are black when they should be white in television receivers because of phase; radio beacons guide aircraft in the skies because the designer took account of phase among other things; and so the examples pile up. Phase, however, is not a subject which you can grasp in one lesson; you will understand it better and better with each succeeding lesson.

LOOKING AHEAD

You have now finished your study of the basic facts about resistors and coils. When you complete a similar study of condensers in the next lesson, you will have secured a thorough basic knowledge of these three important radio parts.

Lesson Questions

Be sure to number your Answer Sheet 6FR-3.

Place your Student Number on every Answer Sheet.

Send in your set of answers for this lesson immediately after you finish them, as instructed in the Study Schedule. This will give you the greaten possible benefit from our speedy personal grading service.

- 1. What effect does an increase in the number of ampere-turns have on the amount of magnetic flux produced by a coil?
- 2. Copy the following list on your Answer Sheet, and indicate after each material whether it is magnetic or non-magnetic.

(a)	Copper.	Exa	mples	(do	not	copy	these): ·	
(b)	Sheet steel.		-	-				
(a)	Dimmond	()	D		7	7	· · · · · · · · · ·	

- (c) Plywood. (x) Brass. Non-magnetic. (d) Aluminum. (y) Air. Non-magnetic.
- (e) Cast iron. (z) Alnico alloy. Magnetic.
- 3. When a pivoted iron object takes the position which gives minimum reluctance to the magnetic circuit of a coil, will we get maximum flux or minimum flux?
- 4. How many flux linkages do we have when 25,000 magnetic lines of force pass through a 3-turn coil?
- 5. Will we get an induced voltage if we suddenly reduce the number of flux linkages in a coil?
- 6. Is any voltage induced in a coil when the flux linkages are not changing?
- 7. Why do we get an induced voltage in a current-carrying coil when we change the reluctance of the magnetic circuit of the coil?
- 8. If a coil has an inductance of 2 henrys, what is its inductance in millihenrys?
- 9. Is the inductive reactance of a coil the same at all frequencies?
- 10. How many degrees out of phase with each other are the voltage and current in a perfect coil?

SINCERE APPRECIATION PAYS

Have you ever watched a dog respond to a friendly pat as a reward for obedience? Have you noticed how a child glows with joy when praised for good behavior? Have you ever felt your own brain cells respond with increased effort when you praise them by saying, "That's a fine piece of work, even if I did do it myself!"?

Yes, everyone responds to sincere and merited praise. It is a tonic to both giver and receiver. It brings greater praise and appreciation back to you. It costs nothing more than a smile and a few sincere words, but it can truly achieve miracles in happiness and success, and put real money in your pocket.

Here are a few ways in which praise can speed up your radio career. Tell the family how much you appreciate their thoughtfulness in being quiet and not interrupting while you study. Make your radio servicing customers feel that you appreciate their business. Find ways to praise sincerely the business men, club leaders and public officials in your community.

Time spent in figuring how to give sincere and deserved praise is well worth while. Let people know that you appreciate *their* fine work, and watch the breaks come your way.

J. E. SMITH



7FR-3

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STUDY SCHEDULE NO. 7

For each study step, read the assigned pages first at your usual speed. Reread slowly one or more times. Finish with one quick reading to fix the important facts firmly in your mind, then answer the Lesson Questions for that step. Study each other step in this same way.

□ 4. How a Condenser Works with A. C. Pages 21-24 This short section describes an important but little understood action—how a. c. can "flow" through a condenser. Read this several times carefully. Answer Lesson Questions 5 and 6.

□ 5. A Simple Condenser-Resistor Circuit......Pages 24-30 When a condenser and resistor are connected together, several important actions occur. The condenser takes longer to charge and, with a. c., a phase shift occurs. Here is more about vectors. Answer Lesson Question 7.

□ 6. Voltage Division with R, L and C.....Pages 30-36 Here is the "heart" of the lesson. You are introduced to some of the basic circuits you will find in every radio device and find out how condenser defects upset them. You will get more details later, but you should study and restudy this section. Answer Lesson Questions 8, 9, and 10.

7. Mail your Answers for this Lesson to N. R. I. for Grading.

8. Start Studying the Next Lesson.

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World Radio History

RADIO CONDENSERS AND HOW THEY WORK

How a Condenser Stores a Charge

ONDENSERS are one of the four corner-stones of radio. Like coils. resistors and tubes, condensers are found in every radio-even the tiniest. In fact, there are usually more condensers than there are of any other type of part. These condensers (acting in combination with other parts) are used to tune the set, to furnish the right power for operating the various sections, to keep the currents in the proper paths, and to perform many other important tasks which must be done for the radio to operate properly. You'll be meeting these useful devices all through your N.R.I. Course, but we should first study the types and basic actions of condensers before we take up their uses in combination with other parts. You'll want to learn about the defects occurring in condensers, as they cause more radio breakdowns than any other part, except possibly tubes. Hence, this lesson gives practical facts about condensers and their troubles

What a Condenser Is. A condenser* is a device for storing electricity. Any condenser, no matter what type it is, consists essentially of two or more conducting surfaces separated by an insulator. The conducting surfaces are usually called plates, and the insulator is known as a "dielectric" (pronounced die-eh-LECK-trick). ► As you'll learn in this lesson, there are a great many types of condensers. They differ in construction, in kinds and shapes of plates, and in the dielectrics used, but basically they all work the same way to store electricity. So before taking up the various types and their uses, let's see just how a condenser works.

CHARGING A CONDENSER

Let's first take the simple case of a condenser which has absolutely nothing—a perfect vacuum—between its plates. When such a condenser is connected to a battery, as in Fig. 1, the battery voltage makes electrons rush from the negative terminal of the battery through wire a to plate A. There the electrons stop, because they cannot flow through the vacuum which separates them from plate B.*

But although these electrons cannot move across the vacuum, their *effect* can. As you learned in an earlier lesson, electrons repel one another—that is, they try to drive one another away. This repelling action occurs even when the electrons are a considerable distance apart, and are separated by a perfect vacuum.

Thus, the electrons which have collected on plate A repel an equal number of electrons from plate B. The electrons from plate B then flow into

^{*} A condenser does not "condense" anything. This is just the popular name for the part, which is more correctly called a capacitor by radio engineers. Similarly, capacitance is more proper than capacity. However, radio men have used these terms so long that they are accepted as correct.

^{*} The electrons normally do not have sufficient energy to escape from the plate and jump the gap. Later you will learn that if the voltage is too high, they can jump the gap between the plates.

wire b, and through the wire to the positive terminal of the battery.

We thus have an electron flow, or current, from the battery to plate A, and an equal current from plate B to the battery but no flow between plates A and B. The flow of electrons onto plate A makes it negative, because it has an excess of electrons. The flow of electrons away from plate B makes this plate positive, because it has a scarcity of electrons. Thus, a difference in potential, or a voltage, exists between plates A and B.

The electron flow continues until plate A is just as negative as the negative battery terminal, and plate B is just as positive as the positive battery terminal. As electrons do not flow across the gap, there can be no continuous flow of current. The electrons just flow long enough to make the condenser charge up so the condenser voltage equals the battery voltage.

Notice the polarity of this voltage, shown in Fig. 2. As you go around the circuit, you find the condenser voltage opposes (bucks) the battery voltage. Therefore, when the condenser voltage is equal to the battery voltage, it cancels the battery voltage and the flow of electrons stops. That is, after a condenser is charged, it does not act like a load-it resembles a source of voltage. Hence, we have the effect of two sources of voltage connected so as to buck each other out, leaving no voltage to maintain electron flow. When the condenser voltage is equal to the voltage of the battery, we say the condenser is fully charged.

The Charged Condenser. Many people believe that a charged condenser contains more electrons than an uncharged one. But what you have just learned shows that this idea is not accurate. It is true that the negative plate has many more electrons

than normal on it; however, each of these extra electrons has forced another electron out of the positive plate. so the total number of electrons on the condenser plates is unchanged. In charging a condenser, we increase the number on one plate and decrease those on the other-the same as if we transferred some electrons from one plate to the other around the circuit. The number of electrons added to the negative plate (and, of course, subtracted from the positive plate) is called the "charge" on the condenser. You may wonder what good it does to charge a condenser, since we don't change the total number of electrons on it by so doing. Well-suppose we disconnect the condenser from the battery. The electrons on the negative plate must remain there, because they have no other place to go. Thus we have an unbalanced condition inside the condenser, with more electrons than normal on one plate and fewer than normal on the other.

As you know, nature always tries to correct an unbalance of this sort. If we provide a path, the extra electrons on the negative plate will rush to the positive plate. In other words, if we connect the plates of our charged condenser with a wire, a current will flow through the wire. (Of course, it will flow only for the time necessary for all the excess electrons to reach the positive plate).

Thus, you might consider a condenser to be a kind of storage battery, or an electrical reservoir. Within limits, we can put electricity into a condenser whenever we wish, and draw it out again whenever we wish. This makes a condenser a highly useful device in radio circuits.

Amount of Charge. The charge on any particular condenser—which, as you just saw, is the number of electrons moved around the circuit—depends on the voltage of the battery used to charge it. The higher the battery voltage, the more the charge will be. However, the voltage alone does not determine the amount of charge we can store in a condenser. The *capacity* of the condenser is just as important as the charging voltage. Let's see what is meant by this term "capacity."

CAPACITY

If we divide the amount of charge on any condenser by the voltage necessary to give it this charge, the result shows us the amount of charge that can be stored on that condenser by an applied voltage of one volt. This is called the "capacity" (sometimes "capacitance") of the condenser. We can express capacity by the equation C = Q/V, where C is the capacity of the condenser, Q is the charge it takes, and V the charging voltage.



FIG. 1. When a condenser is connected to a battery, electrons collect on the negative plate, driving others off the positive plate as shown here.

The capacity of a condenser is really a measure of its ability to store electricity. A condenser with high capacity can store a great deal of electricity for a given voltage, while less can be stored in a condenser of lower capacity. For this reason, its capacity is usually the first thing a radio man wants to know about any condenser he uses. Units of Capacity. Just as ohms are the units of resistance and henrys the units of inductance, so farads (pronounced FAIR-ads) are the units of capacity. A condenser which stores 6.3 million million more elec-



FIG. 2. No current flows when the condenser is charged to a voltage equal to the battery voltage. The condenser voltage then acts as a "bucking" voltage by being equal and opposite to the battery voltage as you progress around the circuit.

trons (which is called a "coulomb" of charge) on its negative plate than on the positive plate when one volt is applied to it has a capacity of one farad. Of course, a farad is a very large unit, so large in fact, that it is not practical for radio work. Instead, two smaller units are used:

1. The microfarad, equal to one millionth of a farad. It is abbreviated $\mu f.$, m f., or m f d.

2. The micro-microfarad, equal to one millionth of a microfarad. Its abbreviations are $\mu\mu f.$, mmf., or mmfd.

All six abbreviations are frequently used in radio, so you should learn them all. The sign μ is the Greek letter "mu" (pronounced MEW).

Sometimes it will be necessary for you to change a value in microfarads to its equivalent value in micromicrofarads, and vice versa. Learn these two simple rules, and you'll find it easy to make the change:

Rule 1. To change mfd. to mmfd., move the decimal point six places to the right.

MFD

Rule 2. To change mmfd. to mfd., move the decimal point six places to the left. For example, suppose we have a capacity of .001 mfd. and we want to know its value in micro-microfarads. We can write .001 as .001000. Then, moving the decimal point six places to the right (Rule 1) gives us 001000 mmfd., or 1000 mmfd. Similarly,-to express 250 mmfd. in microfarads, we would first write 250 as 000250, then move the decimal point six places to the left (Rule 2). This gives us .000250, or .00025 mfd.

Incidentally, radio men usually pronounce .001 mfd. as point double oh one microfarad, or as point zero zero one microfarad. They would pronounce .00025 mfd. as point triple oh two five microfarad, or as point triple zero two five microfarad.

WHAT DETERMINES CAPACITY

Many times in your future radio work, you will use condensers whose capacities can be changed. So you will understand how and why such condensers work, let's now take up the important subject of what determines the capacity of a condenser.

The ability to store a charge depends on the amount of repelling effect we can develop between the condenser plates, so we must either increase the number of repelling electrons or increase the repelling effect of the electrons to store a greater charge. It has been found that the capacity of any condenser depends on just four things —the *area* of the plates, the *number* of plates, the *spacing* of (distance between) the plates, and the *dielectric* used. Let's see just how each factor affects capacity.

1. Area. Besides acting across space to repel electrons from the positive plate, the electrons collected on the negative plate repel each other. For a particular voltage, only a certain number can be made to collect on a plate of fixed size, because they refuse to crowd too closely together. Increasing the area of the plates allows more "standing room" for electrons on the negative plate, so a greater charge is stored for the same voltage. If we double the overlapping area of the plates of a condenser, we double its capacity; if we triple the overlapping area, we triple the capacity; and so on. Notice that it is the overlapping area which is importantany section of the negative plate of a condenser which is not directly over a section of the positive plate does not greatly affect the capacity. You will see the practical use of this fact when we come to variable condensers a little farther along in this lesson.

2. Number of Plates. Suppose we have three plates instead of two, ar-



FIG. 3. Adding plates increa	ases capacity.
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ranged as shown in Fig. 3A. There will be a certain capacity formed between plates A_1 and B, and another capacity between A_2 and the other side of B. As A_1 and A_2 are the same size and connected together, they act like a single plate twice the size of either alone. Since both sides of plate B are used, the effective area is also twice that of one side of B. We thus have twice the effective plate area by adding one more plate. As you just learned, doubling the area doubles the capacity.

The capacity can be further increased by adding more plates. In Fig. 3B, we have capacities between A_1 and B_1 , B_1 and A_2 , A_2 and B_2 , and B_2 and A_3 . Here we have four times the capacity that would be formed by one pair of the same size plates. Thus, the total capacity is equal to the capacity of one pair of plates, multiplied by a number which is one less than the number of plates. For example, if a condenser has a capacity of 10 mmfd. between each pair of plates and has 7 plates, the capacity will be 10 times 6 (7 plates minus one) or 60 mmfd.

3. Plate Spacing. The distance over which the repelling effect must work is important. The smaller the distance, the stronger the force between the plates. This makes it easier to overcome the desire of the electrons to remain on the positive plate, so more are forced away, permitting more to collect on the negative plate. Thus, varying the plate spacing will affect the amount of charge per volt, and so vary the capacity. If we make the distance between the plates of a condenser only half what it was, we double the capacity; if we make the distance a third, we triple the capacity; and so on. This fact, too, has a practical use, as you will find when we treat trimmer condensers in a later section of this lesson.

4. Dielectric. So far, we have been talking about a condenser which has a perfect vacuum between its plates that is, a condenser which uses a vacuum as a dielectric. If, instead, we used mica as a dielectric between the plates, we would increase the capacity as much as 6 to 8 times. Other substances would give different capacity increases.

Since most condensers use some dielectric other than a vacuum, and since a great many of your service problems will be caused by dielectric failures, let's see just what effects dielectrics have on condensers.

EFFECTS OF A DIELECTRIC

You recall that electrons flowing onto the negative plate of our vacuum dielectric condenser repelled electrons from the positive plate and thus allowed the condenser to become charged. What happens to this repelling effect when we put a solid dielectric between the plates?

The two pictures in Fig. 4 show the answer. A dielectric, as you know, is an insulator. Like everything else in the world, it is made up of atoms, which consist of electrons whirling around positive charges. Fig. 4A shows an atom which has a positive charge at its center and one electron



FIG. 4. Electrons in the dielectric are bound to their atoms. They do not "flow"; instead their paths are shifted so they come closer to the positive plate and thus transfer the effect of the extra electrons gathered on the negative plate.

revolving around it in a circle. (Actually, all but one of the 93 known atoms have more than one whirling electron, but for simplicity we'll assume we are dealing with an atom which has only one.)

As you learned in an earlier lesson, a conductor has a large number of free electrons, which can move easily when even a low voltage is impressed across the conductor. But this is not true of an *insulator*. When a voltage is impressed across an insulator, nearly all the electrons within the insulator tend to keep whirling around their central positive charges, because they are bound tightly within their atoms; normally, they will break away only when acted on by a very high voltage. For this reason, the electrons in an insulator are called "bound" electrons. \checkmark

Now, let's close the switch in the circuit shown in Fig. 4B. Electrons start to pile up on the negative plate of the condenser. These electrons repel the bound electrons of the dielectric. Unless this repelling force is extremely large, the bound electrons do not break away from their atoms; however, the force does change their paths within their atoms from the circle shown in Fig. 4A to the oval shown in Fig. 4B. This brings the bound electrons closer to the positive plate of the condenser; they then exert a repelling effect of their own on the free electrons of plate B, driving them out of the plate into the battery.

In this manner, the repelling effect of the electrons on the negative condenser plate is relayed right through the dielectric. Placing the dielectric material between the plates provides bound electrons within the space, and so provides a better transfer medium. This increases the capacity.

Dielectric Constant. You learned a moment ago that any other dielectric gives a condenser a higher capacity than does a vacuum. However, experiments show that the increase in capacity is very slight when air or any other gas under normal pressure is used as the dielectric. In fact, the difference can be detected only by the finest laboratory equipment, so for all practical purposes we can say that a condenser using air as a dielectric has the same capacity it would have with a vacuum dielectric.

If we divide the capacity of a condenser using any given dielectric by the capacity of the *same* condenser using air as a dielectric, we get a number which shows how much the dielectric increases capacity. This number is called the *dielectric constant* * of the dielectric material.

You can readily see that condensers using dielectrics which have a-high dielectric constant have a higher capacity than condensers using materials with a low dielectric constant. The dielectric constants of various common materials used in condensers are: air = 1; paper = 1.5 to 3; paraffin = 2 to 3; mineral oil = 2.5; rubber = 2to 4; mica = 4 to 8; glass = 4 to 10; castor oil = 4.7; porcelain = 5 to 7. Ceramic materials now coming into use have dielectric constants as high as 1500. Thus a condenser which has a capacity of 10 mmfd. using air as a dielectric would have a capacity between 40 and 80 mmfd. with a mica dielectric, and as much as 15,000 mmfd. with one of the new ceramic dielectrics.

VOLTAGE RATINGS

We have shown how increasing the voltage applied to a condenser will increase the charge stored on it. However, there are limits to the amount of voltage we can safely apply to the condenser. For example, when the voltage gets too high for the plate spacing of an air dielectric condenser, a spark will jump between the plates. Thus, the amount we can increase the capacity of an air condenser by decreasing the distance between its plates is limited by the voltage to be applied to the condenser.

The same thing is true for condensers using other dielectrics. If we apply too high a voltage, the bound electrons of the dielectric escape from

^{*} Scientists sometimes call the dielectric constant the "specific inductive capacity" of the material; both names mean the same thing. Don't confuse this term with inductance, a property of coils, as there is no relationship.

their atoms and start a flow of current through the condenser. This burns a hole right through the dielectric. When this happens, we say the condenser "breaks down"; the voltage which is just high enough to cause the condenser to break down is called the "breakdown voltage" rating of the condenser. The value of breakdown voltage for any given condenser depends on the dielectric used in it and the dielectric thickness.

Working Voltage. Condenser manufacturers almost always specify the "working voltage" of a condenser. This is the maximum voltage that can safely be applied to the condenser for long periods of time. Working voltage ratings for solid dielectric condensers, such as paper or mica, are always less than half the breakdown voltage. That is, over twice the working voltage can be applied to solid dielectric condensers before they will break down.

This rating does not mean you have to apply the working voltage to the condenser—this is just the maximum that can be applied safely. Thus, you can use a 400-volt condenser in a 10volt circuit, or in any circuit having not more than 400 volts applied to it.

If a condenser is used in a d.c. circuit, we can put a voltmeter across it to make sure the working voltage is not being exceeded. But, as you have already learned, an a.c. voltmeter used on an a.c. circuit reads effective voltage; this meter reading must be multiplied by 1.4 (approximately) to get the actual peak voltage in the circuit. For example, if the meter shows that 110 volts a.c. is across the condenser, you would multiply this value by 1.4 to find the peak Since 1.4 times 110 equals voltage. 154 volts, the condenser would have to have a working voltage of at least

154 volts to be used in the circuit. Actually, a radio man would use at least a 200-volt condenser, and possibly a 400- or 600-volt type, as these are easily obtained standard sizes and allow an extra amount of safety factor. ▶ Some solid dielectric condensers also have a "peak voltage" rating. This is twice the working voltage; it is therefore just under the breakdown voltage, and represents the maximum voltage the condenser can stand for short periods of time.

Condenser Life. When the condenser has a solid dielectric, like paper or mica, a breakdown ruins the condenser, because a conductive "hole" has burned through the dielectric. Liquid dielectrics are not damaged unless the breakdown is too frequent, in which case the liquid may carbonize and become conductive.

Even when the safe working voltage limits are never exceeded, condensers using solid dielectrics will not last forever. The constant voltage stress on the bound electrons in the dielectric finally forces them out of their atoms and causes breakdown, just as constant bending of a bar of iron will eventually break the iron. Good solid dielectric condensers, used at or below their working voltage, should last from 10,000 to 20,000 working hours—usually much longer than a radio receiver will last.

Condensers must be kept cool to have such long service life. Exposing the condenser to heat speeds up the breakdown process, because bound electrons escape much more readily from their atoms when heated. For this reason, designers usually position condensers in radio equipment so they will not be exposed to too much heat. That brings up a practical hint for your service work—if possible, always put a replacement condenser in the same place the designer put the

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original, or in an equally cool place.

Air condensers will last almost indefinitely. Unlike condensers using solid dielectrics, air condensers are not particularly harmed by breakdown, because the air dielectric is "self-healing" (that is, more air-im-mediately rushes in to replace the air broken down by the voltage). However, they should always be kept dry to minimize the number of breakdowns suffered because, although the condenser itself is usually not injured by breakdown, other parts in the circuit may be harmed by the <u>sudden flow</u> of excess current.

Typical Radio Condensers

So far, we have learned that the condenser is basically a means of storing electricity; that the condenser capacity depends on the area, number and spacing of the plates as well as the dielectric, and that a condenser can withstand just so much voltage before breaking down. Now let us learn something about the types of condensers used in radio, so you can recognize them when you meet them in receivers. Let's also study their weaknesses, so you can see just how and why they become defective, as they so often do.

► Condensers whose capacities cannot be changed are called "fixed" condensers. They are usually classified according to their dielectrics. Thus there are paper, mica, oil, gas-filled, cellulose acetate, polystyrene, electrolytic (in these a chemical deposit is the dielectric), ceramic, rubber and even glass fixed condensers. They are usually in some kind of case, which might be a metal, paper or cardboard box or cylinder, or a molded plastic covering.

▶ Radio men also use two types of condensers whose capacities can be changed — "variable" and "adjustable." Variable condensers—for example, the tuning condenser of a radio —almost always use air as a dielectric. Their name comes from the fact that their capacities can be varied easily. usually just by turning a control knob.

Adjustable condensers are usually used to make corrections or minor adjustments in the capacity of other condensers. For this reason, they are often called "trimmer" or "padder" condensers. They usually have air, or air and some other insulator, as their dielectrics. Their capacities are generally varied by changing the separation of the plates, usually by turning a screw. When they have been adjusted to the desired capacities, they are ordinarily left alone.

Let us now go on to some of the more common types, to see how they are put together and what they look like.

PAPER CONDENSERS

Fig. 5 shows you the important steps in making typical paper condensers. When the condenser is in the form shown in Fig. 5C, it is thoroughly dried in a vacuum to remove air and moisture, then impregnated with wax to keep moisture and air out. Finally, the unit is housed in a cover of some kind—often waxed cardboard a cylinder. Condensers so housed are called "cartridge" condensers because of their appearance. Many paper condensers, especially the larger ones, are housed in metal cases.

If the condenser leads are merely clamped to the foils instead of being



FIG. 5. How paper condensers are made. Waxed paper sheets separate the plates, which are tin or aluminum-foil sheets. After the condenser is rolled up, the projecting ends of the plates are crimped around the spirals formed by the connecting wires. The wires are soldered to tin-foil plates but cannot be soldered to aluminum-foil plates. Instead, the wire spiral is filled with solder and is pressed tightly into the foil. This mechanical joint is held only by wax and the container, so frequently comes apart.

soldered, heating of the condenser when in use may make one lead separate from its foil. This will, of course, disconnect the condenser from the circuit. Such an "open" may occur intermittently or permanently; either condition is something you will meet quite frequently in your servicing. If you service a radio which gives alternately good and bad reception, remember that an intermittent open in a condenser may be the cause.

Voltage Rating. The voltage a paper condenser can withstand depends on the thickness of the waxed paper between the foil sheets. Of course: a thicker dielectric gives a higher voltage rating but also separates the foil sheets more, giving less capacity per unit length of foil. A condenser with a higher voltage rating but the same capacity as another will be larger physically, as longer sheets of foil and paper are needed to give the same capacity because, for a fixed dielectric thickness, the capacity is determined by the plate area, which depends on the length and width of the foil sheets used.

Standard radio receiver condensers of the paper type have ratings of 200, 400, 600 and 1000 volts, with capacities between .0005 mfd. and 2 mfds. Transmitting condensers have much higher voltage ratings.

Inductive and Non-Inductive Types. The method of rolling up the condenser shown in Fig. 5, with all of one edge of each foil sheet fastened to a lead, is known as a "non-inductive" winding. Electrons coming through each wire can move almost directly to any point on the corresponding foil sheet.




Fig. 6 shows an "inductive" winding. In this condenser, the leads are soft metal ribbons, projecting from each foil sheet. When electrons enter the condenser through a ribbon, they have to flow around and around inside the condenser to get to the other end of the plate. This makes the condenser have an inductive effect like a coil. which is often undesirable in highfrequency radio circuits. For this reason, these condensers are found only in power supply and audio circuits, and then only where the convenience of having both leads come from the same end of the condenser is an important factor.



FIG. 6. The "inductive" winding shown here is rarely found in radio, as it can be used only in power packs and audio amplifiers. The noninductive type shown in Fig. 5 can be used throughout radio receivers or transmitters, so is more commonly found.

Condenser Casings. Fig. 7A shows the final appearance of the condenser constructed as in Fig. 5. It is cased in a waxed cardboard tube, with the pigtail leads coming out each end through eyelets in a round cardboard or fiber disc (or through a plug of hard wax).

The condenser may also be cased in a square or rectangular metal box like those shown in Figs. 7B and 7C. The leads extend through the housing but are insulated from it. These condensers are usually provided with mounting brackets, so they can be bolted, riveted ar soldered to the radio chassis.

High-voltage paper condensers which can stand 1000 to 2000 volts (and even 50,000 volts in some special types) are made by using several strips of high-grade waxed linen paper as a dielectric between strips of aluminum foil. The unit is mounted in a container filled with insulating oil. The terminal leads are brought out through porcelain stand-off insulators. Such a condenser is shown in Fig. 7D.

MICA CONDENSERS

Mica is a far better dielectric than waxed paper where very high-frequency currents are involved, as in the r.f. signal circuits of receivers and transmitters. Mica condensers are considerably bigger than paper condensers of equal capacity, so are usually used only where low capacities are required. Mica condensers are generally made in capacities ranging from 10 to 10,000 mmfds. Larger units are available, but are quite expensive.

Fig. 8 shows how a fixed mica condenser is assembled. Thin, clear sheets of mica (only a few thousandths of an inch thick) are stacked in between sheets of copper, lead or aluminum foil. As in the paper condenser, these sheets of foil form the condenser plates. When the desired number of plates have been stacked up, a thicker piece of mica is placed at top and bottom for extra strength, the foil ends are bent over at each end, and a combination lug and clamp is squeezed over each end while the unit is held under pressure. Bakelite is then molded over the unit, leaving only the terminal lugs exposed, as shown in the bottom picture of Fig. 8. This gives the condenser a neat, water-proof insulated housing. Frequently the condenser is then impregnated with wax to seal any tiny cracks which may have formed in the bakelite.

In this condenser the plate area is kept small, but a number of plates are used. This method of manufacture has the advantage of making all except the two outside plates do double duty.



FIG. 7. Here are pictured some of the many condensers you will find in radio apparatus. A = Tubular paper condenser.

- B and C = Metal housing used for moisture and heat protection.
- D = High-voltage type used in transmitters.
- E and F = Postage-stamp size mica condensers.
- G = Larger size having higher capacity.
- H = Mounting holes are provided here so condenser can be rigidly fastened.
- 1 and J = Wet electrolytics; single and triple units.K, L, M, N = Typical dry electrolytics, differing only in the number of units and leadarrangement.
- O and P = Plug-in dry electrolytics. Q, R = Top and bottom views of a typical trimmer.
- S = Dual trimmer unit.
- , U, V = Variable condensers differing in size and number of plates.
- T, U, V = Variable condenser. W = A three-gang variable condenser.

In the condenser shown in Fig. 8, plates 2 and 1 are really one condenser, and plates 2 and 3 are another. Since plates 1 and 3 project out the same side and are connected by the lug clamp, this gives the whole condenser a capacity equal to twice the capacity of each individual condenser. Therefore, using only *three* plates, we have a capacity double what we would have by using *two* plates of the same size and separation.

When mica condensers are made



FIG. 8. Mica condensers are made of alternate layers of mica and metal plates, as shown at the top. The leads are clamped over the foil projections, then the unit is molded in a bakelite housing.

with many plates, all the foils sticking out at each end are usually soldered together, instead of just being held with a clamp. This makes a better electrical connection.

Figs. 7E, F. G and H show some typical fixed mica condensers in their bakelite housings. The terminals of these condensers are strong enough to support the condensers without need

for other mounts. When the condenser is to be bolted to the chassis or when several condensers are to be bolted together, units like that in Fig. 7H, with mounting holes molded in the bakelite housing, are used.

▶ Since mica is a very good insulator, mica condensers have amazingly high working voltages for their size. Receiver types are all 500 to 600 volts, while transmitters use slightly larger sizes rated as high as 2500 volts.

ELECTROLYTIC CONDENSERS

The electrolytic condenser shown in Fig. 9 consists of an aluminum rod placed in a metal container filled with a conductive chemical solution. The solution is called an "electrolyte" (pronounced *e-LECK-troh-light*). The rod is called the "anode," which is the general term for the positive part or terminal of any device.

To make the condenser, the positive terminal of a d.c. voltage source is connected to the anode, and the negative source terminal to the metal container. In a few hours, electrochemical action produces a thin, high-resistance film of aluminum oxide on the anode.



FIG. 9. Electrolytic condensers depend on the formation of an aluminum oxide film on the anode as the dielectric, then the anode rod and the electrolyte form the "plates" of the condenser. The can is called the cathode, as it is the means of making connection to the electrolyte which is the true negative plate.

The condenser is then said to be "formed," and is ready for use.

The anode is one plate of the finished condenser, the aluminum oxide film is the dielectric, and the surface of the electrolyte which is in contact with the dielectric film is the other plate. The metal container acts only as a container and as a lead from the outside circuit to the electrolyte; it is not a plate of the condenser. (The container is usually called the "cathode," which is the general name for the negative part or terminal of an electrical device.)

▶ Using a rod as the anode limits the capacity, as the area of the rod surface is limited. More modern con-

FIG. 10. Top and side views of a typical wet elec-

trolytic condenser. A = Anode, a plain or etched aluminum foil sheet folded in a zig - zag or crimped manner and riveted

to a support rod. B = Cathode, an aluminum can.

C = Aluminum cover. D = Semi - porous gasket under cover, which prevents leakage of liquid electrolyte yet allows gases to escape.

E = Level of electrolyte. F = Insulating material, which prevents short circuits between anode and cathode.

G = Insulating gasket which separates anode terminal from cathode and seals container at this point.

H = Connection to anode.

densers obtain greater surface areas. A cross-section of a typical electrolytic condenser is shown in Fig. 10. Here the anode is a sheet instead of a rod, and is folded back and forth within the container. This increases the surface area of the anode, and so gives higher capacity. In this particular condenser, the anode is "crimped" so it will take up as little space as possible; in other types, the anode is often wound in a spiral for the same reason. Frequently the anode is also chemically etched so that it has a rough surface, and therefore more surface area.

The thin dielectric and large plate area allow an electrolytic condenser to be physically rather small and yet have a very high capacity. In fact,



Courtesy Sprague Products Company electrolytics (as radio men call them) can have a much higher capacity for a given size than any other kind of condenser, with the possible exception of the new ceramic type. (As an example of comparative sizes, an electrolytic condenser of about the same dimensions as a 2-mfd. paper condenser might have a capacity of around 60 mfd.) This fact, and their relatively low cost, are the chief reasons electrolytics are used so much in radio circuits.

► Electrolytics which have a fluid electrolyte are often called "wet" electrolytics to distinguish them from two other types—semi-dry and dry electrolytics. Semi-dry electrolytics are like wet electrolytics except that a thickening material is added to the electrolyte, making it jelly-like and almost spill-proof.

The dry electrolytic condenser is today taking the place of the semi-dry type. Fig. 11 shows how dry electrolytics are made. Here the cathode is a sheet of pure aluminum, while the anode is a sheet of aluminum which



FIG. 11. How dry electrolytics are made.

has been "formed," so that it is covered with a thin dielectric film. The two electrodes are separated by strips of cheesecloth, paper or cellophane, which have been filled with an electrolyte in paste form. The four strips of material are rolled into a compact cylinder, flexible wire leads are attached to each electrode, the unit is mounted in a cardboard or other container, and the whole condenser is dipped in hot wax several times to make it air-tight. Sometimes dry electrolytics are sealed in metal cans similar to the containers for wet types.

Capacity of Electrolytics. The thickness of the dielectric film of an electrolytic depends on the voltage used to form it. High forming voltages give fairly thick dielectric films, while low voltages give thinner films. As you have already learned, the thinner the dielectric the closer the plate spacing and the higher the capacity will be. So, other things being equal, the electrolytic formed at a low voltage will have a higher capacity than one formed at a high voltage.

The forming voltage also determines the breakdown voltage of the condenser, since a voltage higher than that used to form the dielectric film cannot be applied to it without breaking the dielectric down. On the other hand, the ability of the condenser to maintain the dielectric film depends on the applied d.c. voltage. That is, the dielectric film becomes thinner as the applied voltage is decreased, so the capacity and working voltage ratings are not reliable unless the applied voltage is near the forming voltage. Hence, the working voltage rating of an electrolytic is usually about 90% of the forming voltage. Notice this means the breakdown voltage is only slightly higher than the working voltage. The electrolytic condenser thus differs from those using solid dielectrics in the relationship of its voltage ratings. These condensers should be used in circuits having voltages close to their working voltage ratings, although they can be used on lower voltages if the increased capacity will not matter.

The highest standard working voltage for electrolytic condensers is 450 to 475 volts. These condensers will break down if the applied voltage exceeds about 525 volts. (One manufacturer has produced a condenser with a breakdown rating of about 600 volts.) Capacities up to 60 mfds. are available in the 450-volt electrolytics, and capacities as large as 125 mfds. may be found in condensers with working voltages around 50 volts.

Service Notes on Electrolytics. Unlike all other common forms of condensers, electrolytics must always be connected into a circuit with the proper polarity. The anode must always be connected to the positive terminal of the circuit, and the cathode must be connected to the negative terminal. Now you see why the terminals are so named. Condensers with solid dielectrics do not have polarity so the leads can be interchanged at will.

Dry electrolytic condenser leads are labeled or colored, so the proper connections can be made. An identifying code or table is stamped on the case of most replacement types. Thus, you may find a red lead is the anode (+)and a black lead is the cathode (-)connection for a single-section dry condenser. When in a cylindrical container, the case may be marked (+)near the positive lead.

The metal container of wet and semi-dry electrolytics is the cathode, and so is connected to the negative terminal of the circuit. The other condenser terminal (usually in the center of the condenser) is the anode, and should go to the positive part of the circuit.

If you connect electrolytics into a circuit with the wrong polarity, the dielectric will break down almost at once. This is usually not too harmful with wet and semi-dry electrolytics, because they are self-healing—that is, the dielectric will repair itself within a short time after the connections are corrected. But dry electrolytics do not have this property; if broken down, they are ruined.

▶ The fact that electrolytics must be connected with proper polarity means they can be used only on d.c. or on d.c. mixed with a c. In the latter case, the d.c. part of the voltage must be larger than the a.c. part, so that the polarity of the applied voltage will never be reversed by the a.c. variations. For example, it is all right to connect an electrolytic with a suitable working voltage to a circuit which furnishes 110 volts d.c. and 40 volts (peak) a.c., provided the positive terminal of the condenser is connected to the positive terminal of the d.c. voltage. Then the voltage applied to the positive terminal of the condenser will vary from 110 minus 40 to 110 plus 40, or from plus 70 to plus 150, but will always be positive, so the condenser will not break down. If the circuit furnishes 40 volts d.c. and 110 volts (peak) a.c., however, the voltage on the positive terminal of the condenser will vary from 40 plus 110 to 40 minus 110, or from plus 150 to minus 70, and the condenser will break down when the voltage reverses polarity.

▶ As you will learn a little later, electrolytics are often used in circuits where d.c. and a.c. are mixed. When vou use one in such a circuit, you must be sure not only that the circuit does not reverse polarity, but that the peak value of the a.c. added to the value of the d.c. does not exceed the working voltage of your condenser. Electrolytics break down very quickly when the working voltage is exceeded. As with breakdowns caused by wrong polarity of connections, a voltage breakdown does not usually cause permanent harm to wet or semi-dry electrolytics, but ruins dry electrolytics. Servicemen usually play safe when replacing condensers by being sure the

replacement has a working voltage rating equal to or greater than the original part.

▶ One disadvantage of electrolytics is that the aluminum oxide film which forms their dielectric is not a true insulator, so some electrons can flow through the condenser. This is called a leakage current, as it "leaks" through the dielectric. Since the electrolyte (through which the leakage current must flow) has some resistance, the flow of leakage current through it creates heat within the condenser This heat drives gas out of the electrolyte. Normally, the vent on the top of wet and semi-dry electrolytics allows this gas to escape; if the vent becomes plugged for any reason, the condenser may blow up.

The escaping gas carries a certain amount of electrolyte with it. In time. this action disturbs the chemical balance within the condenser, increasing the resistance of the electrolyte: also. since the electrolyte forms one plate of the condenser, and lowering its level is the same thing as decreasing the plate area, lowering the electrolyte level reduces the condenser capacity. The added resistance of the electrolyte increases the heating effect of the flow of leakage current, thus speeding up the destructive process. Eventually, this combination of actions ruins the condenser. From 2000 to 10.000 hours of service can be expected from a good wet electrolytic.

▶ Much the same effect occurs in dry and semi-dry electrolytics, in which the heat caused by the flow of leakage current tends to dry out the paste electrolyte. In fact, these types will not usually last as long as wet electrolytics; dry electrolytics are used in preference to wet electrolytics in radio circuits only because they are usually much smaller in size, may be mounted in any position, and cost less. Of course, heat from any source is as bad for electrolytics as the internal heat caused by leakage current. Keep electrolytics away from tubes, transformers and large resistors that may radiate considerable heat.

Identifying Electrolytics. Sometimes it is hard to tell just what type an electrolytic is by looking at it. If the condenser is in a metal can with water-tight gaskets around the termi-



FIG. 12. Two views of a typical trimmer.

nals, and a rubber nipple through which gas can escape, you are usually safe in assuming it is a wet or semidry type. Pick such a condenser up and shake it—if you hear a swish, it is a wet electrolytic. Dry electrolytics usually have flexible leads and are cased in cardboard containers or in sealed metal cans which have no vent to permit gas to escape.

Figs. 7I and J show two kinds of wet electrolytics. (Semi-dry would look exactly the same.) A single unit is shown in I. The triple unit in J has three terminals on the top of the can as the positive terminals, while the can is the common cathode.

Eight common types of dry electrolytics are shown in Figs. 7K through 7P. The terminals of condenser 70 come out to a plug mounted on the end of the condenser. This condenser is connected into the circuit by being plugged into a receptacle. Type 7P is also a plug-in condenser, with flat terminal pins which fit into the special wafer mounting shown beside it. These pins are twisted after the condenser is plugged in; this both makes a good electrical connection and holds the condenser firmly in place.

TRIMMER CONDENSERS

A trimmer condenser is used to make a small adjustment in the capacity of a circuit. A typical trimmer is shown in Fig. 12.

Air and mica are used as dielectrics. One plate is riveted to the bakelite base of the unit, and the other plate. usually made of spring brass or phosphor bronze, is moved close to or away from the fixed plate by turning the adjusting screw. When the two plates are close together, the mica sheet is the dielectric, and the capacity is a maximum. When the two plates are farther apart, both air and mica are between them and the capacity is less. In some condensers, the adjusting screw makes electrical contact with the movable plate: in others it is insulated from the movable plate: it is



FIG. 13. Adjustable condenser symbols used in wiring diagrams. The one at A is used for trimmer condensers. The other two are used for variable tuning condensers usually, although sometimes they are used to indicate trimmers too.

always insulated from the fixed plate. When a trimmer with large capacity is wanted, several pairs of plates are used, with alternate plates connected together.

▶ Trimmer condensers are also called "equalizing," "neutralizing," "aligning," "phasing" or "padding" condensers, depending on how they are used in the radio circuit. Fig. 13A shows



FIG. 14. How a variable air condenser works.

the symbol used in your N.R.I. Course and by many radio manufacturers to indicate a trimmer, while 13B and Cshow the ordinary symbols for a variable condenser, which some authors and manufacturers also use for trimmers.

▶ Typical trimmers are shown in Figs. 7Q through 7S. Because these trimmers open like a book, they are often called "book type" trimmers.

VARIABLE CONDENSERS

The series of pictures in Fig. 14 show the working principle of the variable air condensers used to tune many modern radios. (For clearness, only one set of plates is illustrated, but most such condensers have several sets, as shown in Fig. 15.) The plate at the left in A, B, C, D and E is a fixed plate, called the "stator." The smaller plate is pivoted on a shaft so that it may be rotated; this plate is called the "rotor."

When the plates are in the position shown in A, the condenser has a minimum capacity, because, as you learned earlier in this lesson, only the *overlapping* parts of plates have a capacity between them. As the rotor is turned past the stator, more and more of the plates overlap, as shown by the shaded areas, and the capacity of the condenser increases. When the plates reach position E, the capacity is a maximum. Relatively high-capacity variable condensers are made by using large plates or by adding additional rotors and stators. When this latter is done, the rotor plates are connected together electrically by the shaft to which they are attached, and the stators are electrically connected by a metal bar built into (but insulated from) the condenser frame.

Several typical variable condensers are shown in Figs. 7T through 7W. Those pictured in T, U and V are similar, differing only in size. Unit Wis really three separate condensers with the rotors mounted on the same shaft, so that the capacity of all three condensers is varied at the same time. Each condenser stator is insulated from the other two, and separate connections are made from each to the radio circuit it is used in. This kind of unit, in which the capacities of two or more condensers are varied by turning one shaft, is called a "gang" condenser. The one pictured in W is a three-gang condenser.

Although the condensers in a gang





are carefully made, manufacturing tolerances result in slight differences in capacity. To correct this and make the capacities of condensers in the gang "line up" with each other and with the circuits with which they are used, a trimmer condenser is usually connected to each condenser. The trimmer is adjusted to get exactly the right capacities in each section. Thus, the trimmer is a "corrector" and once adjusted is left alone until some change makes it necessary to repeat the adjustment.

Connecting Condensers Together

Now that you know what condensers are and how they work, let's consider a very practical point—what happens when condensers are connected together. Very frequently, you may need a certain capacity and find you do not have the right value, but have several condensers of other sizes to choose from. What is the effect on capacity when condensers are connected in parallel or in series?

CONDENSERS IN PARALLEL

Going back to Fig. 3A, you will see that the condenser plates A_1 and A_2 are connected together so that the capacity A_1 -B is in parallel with the capacity $B-A_2$. (There is a separate path from one terminal to the other through each condenser.) This parallel connection increases the capacity in this one condenser and would also increase the capacity if these were separate condensers. So if one condenser doesn't have enough capacity for the use to which you want to put it. you can connect another condenser to it in parallel to get a higher capacity. The capacity of a combination of two or more condensers connected in parallel is equal to the sum of the capacities of the individual condensers.

Fig. 16 shows why. Since they are connected in parallel, each condenser has the full line voltage across it. Each stores the amount of charge it usually does with such a voltage across it, so the amount of charge stored is the sum of that in all the condensers. As the <u>capacity</u> is equal to the charge divided by the voltage, this adding of charges means the capacity of the combination is equal to the capacity of all three condensers added together. Thus, the capacity of the combination in Fig. 16 is 33 mmfd. (10 mmfd. plus 11 mmfd. plus 12 mmfd. equals 33 mmfd.).

CONDENSERS IN SERIES

When condensers are connected in series, as in Fig. 17, the capacity of the combination is less than the capacity of the smallest condenser in the group. Since plates b and c in Fig. 17A are connected by a wire, the same charge must be on each of them; similarly plates d and e have equal charges. Hence, we can consider the series combination to be really one condenser with plates a and f and a dielectric equal in thickness to the sum of the thicknesses of the dielectrics of all three condensers (Fig. 17B). Naturally, the greater dielectric thickness makes the capacity of the combination less than that of any one of the individual condensers.

From this you can readily see that if several condensers having equal capacities are connected in series, the total capacity of the group equals the capacity of one condenser divided by the number in the group. For example, if all three condensers in Fig. 17A are identical, then the equivalent condenser shown in Fig. 17B will have the same plate area as any one of the condensers and three times the dielectric thickness. Its capacity will therefore be one-third that of any one of the capacity of one condenser). The total capacity of four identical condensers in series would be C/4, of five in series C/5, and so on.

▶ This method of finding the capacity of a series combination can be applied only to condensers of equal capacity. But there is a simple rule for finding the capacity of any two condensers connected in series, whether they have the same capacity or not; just divide



FIG. 16. Condensers in parallel give more capacity.

the product of the two capacities by the sum of the two capacities. Using symbols:

$$C = \frac{C_1 \times C_2}{C_1 + C_2}$$

where C is the capacity of the series combination, C_1 the capacity of one condenser, C_2 the capacity of the other condenser, with all capacities expressed in mfd. or all in mmfd.

▶ If you have more than two unequal condensers connected in series, find the combined capacity of two of them by this equation. Then treat this combined capacity as C_1 in the equation, take the third condenser as C_2 , and again solve for C. You can keep this up for as many condensers as you have in the series circuit and your final answer will be the combined capacity of all of them. Notice—the combined capacity of condensers in series is found in exactly the same way that you learned to find the combined resistance of resistors in parallel.

VOLTAGE RATINGS FOR CONDENSER COMBINATIONS

When condensers are connected in parallel, each has the source voltage across it. Therefore, each condenser must have a working voltage at least equal to the source voltage.

While each condenser in a series combination has only part of the source voltage applied to it, still the safest thing is to make sure all of them can stand the full source voltage. But sometimes you won't be able to get condensers with high enough working voltages to observe this rule. Let's take a practical problem and see the simple trick you can use to solve it.

Suppose you have a d.c. source with a voltage of 1000 volts which must be used with a 2-mfd. capacity. (This is a problem which often comes up in transmitters and high-voltage power packs.) You have two 4-mfd. paper condensers, which will give the 2-mfd.



FIG. 17. Condensers in series give less capacity.

you want if put in series. But suppose the working voltage rating of each condenser is only 500 volts.

► If you can be sure each condenser will get no more than 500 volts across it, you'll be safe in using them in series. But how can you be sure?

Leakage Current. A perfect condenser (excluding electrolytics) will pass no d.c. However, no solid dielectric condenser is perfect; there are always some free electrons in even the best dielectric and, when a voltage is applied across the condenser, these electrons will flow through the dielectric. (Of course, this flow of current is much smaller than that necessary to cause breakdown.) Further, there is always some flow of current between the leads across the surface of the case of the condenser; in fact, this current



FIG. 18. A practical condenser acts like a perfect condenser in parallel with a high resistance.

is usually considerably higher than the current through the dielectric.

The amount of current that flows from one terminal of the condenser to the other when a d.c. voltage is placed across the condenser is known as the "leakage current." Condensers having impurities in the dielectric, or so made that moisture can get in, will have much higher leakage currents than well-made ones. Condenser manufacturers always test the leakage of their solid dielectric condensers; those which pass too much current are rejected. For most purposes, the leakage current is so small it is ignored altogether.

▶ Where the leakage is important, we can take leakage current into account by considering a practical condenser to be a perfect condenser with a high resistance shunted across it, as shown in the cutaway section in Fig. 18. This way of picturing a practical condenser is often very useful.

The high resistance shunting the condenser is called the "leakage resistance" of the condenser. Naturally, a condenser with a high leakage resistance will pass less d.c. than one with a lower leakage resistance.

▶ Since the perfect condenser won't pass d.c., it must be the current passing

through the leakage resistance which determines the $I \times R$ drop (or voltage) across the condenser.

Now if one of the 4-mfd. condensers has a leakage of 50 megohms and the other has a leakage resistance of 200 megohms, as shown in Fig. 19A, the total resistance is 250 megohms. Using Ohm's Law, you could figure the current flow and resulting voltage division. You would find that the line voltage will divide according to the ratio of the resistances. Thus, the condenser with 50 megohms will get 50/250 or $\frac{1}{5}$ the source voltage. This puts 200 volts (1/5 of 1000) across this condenser and leaves 800 volts across the other condenser. Since the 800 volts across the latter condenser is considerably more than its safe working voltage rating of 500 volts, it will break down fairly soon. Then the full





1000 volts will be applied to the other condenser, breaking it down almost at once.

If the condensers are used in a circuit where a small current flow will not matter, you can solve this problem very easily by putting a 10-megohm resistor across each condenser. These added resistors are in parallel with, and considerably lower in ohmic value

than the leakage resistances of the condensers. As you learned in your lesson on resistors, combining two resistors of widely different values in parallel gives a combined resistance just about equal to the smaller of the two. Therefore, the total resistance of each condenser-resistor combination will be about that of the added resistors, as shown in Fig. 19B. Thus the resistances are made equal, so the voltage will divide about evenly across the combinations, and each condenser will be operating at its working voltage.

The same trick can be used if your condensers are electrolytics, but since electrolytics have much lower leakage

How a Condenser Works with A.C.

So far, we have dealt with the charging action of a condenser when a d.c. voltage is connected across it. You have learned that leakage current is the only current flow through a condenser when a d.c. source is used, and this flow can be ignored in most instances. Ordinarily, when a d.c. voltage is applied to a fixed condenser, the condenser charges up to the source voltage value, as long as the voltage does not exceed the safe working voltage rating of the condenser. When the charge on the condenser has reached its maximum, the movement of electrons in the circuit stops, so all current flow ceases until a different voltage is applied or until the condenser is discharged. 1

The nature of an alternating current provides a different action, however. As you will recall, electrons do not flow always in one direction when acted on by an a.c. voltage. Instead, they move a short distance in one direction in the wire; then, when the voltage reverses, they move in the other direction. This back-and-forth resistances than paper condensers, you'll have to use lower shunting resistors. Between 10,000 ohms and 50,000 ohms will be satisfactory. (If you connect two electrolytics in series. remember that the positive terminal of one must connect to the negative terminal of the other.)

These extra resistors act "bleeders," drawing or "bleeding" an extra current from the source of such value that they determine the voltage division. Naturally, this method of getting an even split of voltage across two condensers can be used only if this extra current is not objectionable in the circuit.

motion of the electrons in the circuit permits us to say that alternating current flows through a condenser. Let's examine this action more closely.

HOW ALTERNATING CURRENT FLOWS THROUGH A CONDENSER

When electrons are moving back and forth in the circuit due to an a.c. source, electrons must flow in and out of the plates of the condenser. When the electrons flow into one plate, the bound electrons in the dielectric are forced away from that plate and force electrons out of the other condenser plate.

When the a.c. voltage reverses, the electrons are forced back into the second plate of the condenser. And now the electrons on the second plate act just the way the electrons on the first plate did a moment before-they push the bound electrons in the dielectric away and the movement of the bound electrons now push the electrons out of the first plate.

This action is shown in Fig. 20. At A, we start with the electrons at rest in the dielectric. Then, as electrons are forced into plate 1, the path for the bound electrons in the dielectric shifts, forcing electrons out of plate 2 as shown at B.

This movement of electrons causes the condenser charge to reach a maximum when the a.c. voltage reaches a maximum on this half cycle. Then, as the a.c. voltage decreases, the electrons redistribute themselves about the circuit, allowing the electrons in the dielectric to return to their neutral position as shown in Fig. 20C. As the cycle reverses, electrons begin to flow into plate 2, forcing a movement in

the bound electron paths in the dielectric when acted on by a.c., just as there is a back-and-forth movement of free electrons in the conductor. Therefore. around the complete circuit there is a back-and-forth motion of electrons and we are perfectly justified in saying that an alternating current passes through a condenser. Scientists call this current within the dielectric a "displacement current." This new name is given the current flow because there is no actual electron movement from plate to dielectric or from dielectric to plate. Instead, there is a displacement of electrons within the dielectric which passes on the current flow action of the entire circuit.



FIG. 20. When a.c. is applied, the bound electrons move first one way, then the other, so, in effect, an alternating current flows through the condenser.

the dielectric in the opposite direction so that electrons are forced out of plate 1 as shown at D. This action reaches a maximum, charging the condenser in this reversed direction. Then the voltage decreases toward zero, allowing the bound electrons to return to the neutral position shown at A, after which the cycle is repeated over and over. Fig 21 shows the points on an a.c. cycle corresponding to this action. The condenser charges between A and B, discharges between B and C, charges with the opposite polarity between C and D, discharges again between D and A, and repeats the cycle over and over.

From the foregoing, you can see that there is a back-and-forth motion of It is important that you get the idea clearly that an alternating current is just a movement back and forth of electrons, and further, this actual amount of movement is over an astonishingly short distance. However, when electrons move in the circuit, it is not the distance they travel that is important, it is the number of electrons which actually do move within a given period of time which gives us a measure of current flow.

CAPACITIVE REACTANCE

Although a condenser permits a flow of alternating current through the circuit, it also restricts the current flow. Work must be done to move the bound electrons in the dielectric, so there is an opposition to the flow of current. Actually, the charge-storing ability will limit the current flow.

Now, the larger the condenser, the greater its ability to store a charge. As the current flow depends on the number of electrons in the charge, we



FIG. 21. The points on the cycle corresponding to the electron path shifts shown in Fig. 20.

see that the larger the capacity, the less the opposition to current flow.

The opposition of a condenser to an alternating current is called reactance, just as it is for a coil. This reactance, as is that of a coil, is measured in ohms. However, notice this important difference—for a condenser, the larger the condenser the less the reactance, while for coils, the larger the inductance the greater the reactance.

Furthermore, frequency changes also have an inverse effect upon condenser reactance. That is, the higher the frequency, the less the reactance of a condenser, whereas with a coil, the higher the frequency the greater the reactance. As you will soon see, these are very important differences.

You can figure out the reactance of a condenser by this simple method:

Multiply the frequency of the current in cycles per second by the capacity of the condenser in microfarads, then divide the number 159,000 by this result.

The equation for this is:

$$X_{\rm C} = \frac{159,000}{f \times C}$$

Where X_6 is the capacitive reactance in ohms, f is the frequency in cycles per second, and C is the capacity in mfd.

Another way we might express this is:

$$X_{\rm C} = \frac{1}{6.28 \times f \times C}$$

where C is in farads. (Compare this with the equation for inductive reactance you learned in your lesson on coils.)

▶ Let's take a few practical examples, to see just what these important equations mean. We will use only the first equation, since its units are more convenient. (The second equation, of course, is exactly the same as the first one; the difference in units is what makes them look different.)

Suppose we have a condenser of 10mfd. capacity fed by a source supplying 100-cycle current. Then:

$$X_{c} = \frac{159,000}{f \times C} = \frac{159,000}{100 \times 10}$$
$$= \frac{159,000}{1,000} = 159 \text{ ohms}$$

so the a.c. reactance of our condenser is 159 ohms. (Remember, this a.c. reactance has nothing to do with the d.c. leakage resistance; a condenser has a reactance only when it is used in an a.c. circuit.)

Now if we supply the same condenser from a source which furnishes 1000-cycle current:

$$X_{\rm c} = \frac{159,000}{f \times C} = \frac{159,000}{1000 \times 10}$$
$$= \frac{159,000}{10,000} = 15.9 \text{ ohms}$$

so *increasing* the frequency of the current *decreases* the reactance of our condenser. Notice that the increase in frequency causes a proportionately equal decrease in reactance; *increasing* the frequency 10 times *decreases* the reactance 10 times. If we keep the frequency at 100 cycles and increase the capacity of the condenser to 100 mfd:

$$X_{\rm C} = \frac{159,000}{f \times C} = \frac{159,000}{100 \times 100}$$
$$= \frac{159,000}{10,000} = 15.9 \text{ ohms}$$

so increasing the capacity of the condenser decreases its reactance. Just as the increase in frequency did in the previous example, the increase in capacity causes a proportionately equal decrease in reactance; increasing the capacity 10 times decreases the reactance 10 times.

▶ Let us sum up what we have just learned in one easily remembered rule: An increase in frequency or an increase in capacity will cause a decrease in the reactance of a condenser.

Of course, the reverse effect is also true; a decrease in frequency or capacity causes an increase in reactance.

Since a drop in reactance means it is easier for a.c. current to pass through the condenser, you can readily see that high-frequency currents pass through a condenser much more easily than do low-frequency currents; also, that a current of any frequency flows more easily through a condenser of high capacity than through one of low capacity. You will see how extremely important these facts are a little later on in this lesson, when we discuss some practical condenser circuits.

A Simple Condenser-Resistor Circuit

Now we know that a condenser will store a charge when a d.c. voltage is applied, and that condensers permit a.c. current to flow in the circuit, limiting the current flow by their reactance. What happens when we connect other parts with condensers? Let's combine a resistance with a condenser and consider the action both for d.c. and a.c. sources. We'll start with a d.c. source.

TIME CONSTANT

With the resistor and condenser connected to a d.c. source, as shown in Fig. 22, a very interesting action occurs. If the condenser were connected by itself directly to the source, the condenser would charge to the full source voltage almost immediately. However, when we put a resistor in the circuit, we find that it takes longer—the charging time increases to as much as several minutes, if the resistor and condenser are large enough. Let's learn more about this action.

The effective voltage in this circuit -that is, the voltage which acts to cause current flow-is equal at any time to the source voltage minus the back, or bucking, voltage of the con-When the circuit is first denser. closed, there is no charge on the condenser and therefore no voltage across it, so the effective voltage is equal to the full source voltage. The current flow in the circuit is then equal to the source voltage divided by the value of As this current flows resistor R. through the circuit, however, the condenser begins to store a charge, and voltage builds up across it. This, of course, makes the effective (or current-moving) voltage less than the full

source voltage, so the current through the resistor and circuit must drop. As the condenser continues to charge, the effective voltage becomes less and less, making the circuit current continually smaller. Eventually, when the condenser voltage equals the source voltage, there is no effective voltage and all current flow stops.

▶ The effect of the resistor is to increase the time necessary to charge the



FIG. 22. The condenser charges through the resistor more slowly than it would if the resistor were not used.

condenser. Even a very large condenser will charge almost at once if there is a sufficient current flow into it, but when, as in this circuit, the current flow is limited to a small initial value by the resistor, and then grows even smaller, it may take a considerable time for enough electrons to flow into the condenser to charge it completely.

The amount of time it takes for the condenser to charge up with a resistance in series with it is an important factor in the action of many radio circuits. It has been found that multiplying the capacity of the condenser in mfd. by the value of the resistor in megohms gives a figure equal to the time in seconds that it takes the condenser voltage to reach about 63% (actually 63.3%) of its final value. This time is called the "time constant" of the condenser-resistor combination. The time it takes a condenser to reach its full voltage is many times this special value, but comparing time constants directly is often a useful way to compare the action of one radio circuit with that of another.

Fig. 23 is a graph showing how the

condenser voltage increases with time. Notice the rapid rise at first, followed by a gradual tapering off until the final condenser voltage is reached a relatively long time later on. As shown, the time constant is the amount of time it takes to reach about 63% of full charge.

You will meet time constants frequently in later lessons, for they have much to do with such important things as the ability of an amplifier to amplify signals without distortion. For the present, just remember that the time constant of a condenser-resistor combination is the time in seconds required to charge the condenser to about 63% of the charging voltage. Further, this time can be obtained directly from the condenser and resistor values by multiplying the capacity of the condenser in mfd. by the ohmic





value of the resistor in megohms. Thus, a 10-mfd. condenser and a 5megohm resistor have a time constant of 10×5 or 50 seconds, or almost a minute to charge to 63% of the source voltage, while a .05-mfd. condenser and a .1-meg (100,000 ohms) resistor will charge to this same value in .05 \times .1, or .005 second, thus, the larger the resistor or the condenser, the longer the time constant.



FIG. 24. How the voltages are divided when d.c. is applied at A. The division for a.c. is shown at B.

ACTION WITH A. C.

You recall that in the lesson on coils, we compared the action of a coil and a lamp in series on a.c. with their action on d.c. Let's make the comparison again, this time substituting a condenser for the coil. Of course, the lamp represents a resistor, so we have the same circuit as Fig. 22. However, the lamp shows when current flows by lighting.

Starting first with a d.c. source, as shown in Fig. 24A, we find that the lamp will not light. This indicates that no d.c. flows through the condenser. Actually, there is an initial flow of current when the circuit is first completed, but this current flows for too short a time to light the lamp even momentarily. When the condenser is charged, the entire d.c. voltage is across the condenser and none across the lamp. The d.c. voltage drops in the circuit check with Kirchhoff's Voltage Law, as 115 volts across the condenser added to zero lamp volts equal the 115-volt source voltage.

Suppose we now change_to a 115volt a.c. source, as in Fig. 24B. With a.c. voltmeters being used now to measure the voltage drops, we find that the two voltage drops add up to 158 volts; which is much higher than the source voltage. Clearly, we cannot apply Kirchhoff's Voltage Law directly to this circuit. Once again we have met the subject of *phase*.

The voltage drops across the condenser and lamp in Fig. 24B do not add up to the a.c. source voltage because these a.c. voltage drops do not reach corresponding peak values at the same instant of time in each cycle. In other words, one a.c. voltage may be at the zero point in its cycle at the instant when the other a.c. voltage is at its maximum or peak value. We encounter this situation whenever we use coils or condensers along with resistors in radio circuits, and radio men say that there is a phase difference between these a.c. voltage drops.

The only time we can add directly the a.c. voltages which a.c. meters measure is when these voltages are in phase with each other (when all reach corresponding peak values in each cycle at the same instant of time, and all drop to zero together). This occurs only when all the parts in the circuit are identical (all are perfect resistors, all are perfect coils or all are perfect condensers).

The a.c. meters do not show the instantaneous voltages; they show the effective values over a period of time. If they could show the instantaneous values, we would find they would add up properly at all times. Instead, we must combine the meter readings in a way which will take phase into account in order to see just what goes on. Let us now see how phase enters into the picture when the a.c. circuit has a -condenser in it.

Phase. You remember that voltage and current are always in phase in a purely resistive a.c. circuit, and that they are always 90° out of phase in a purely inductive a.c. circuit, with the current lagging behind the voltage. In a capacitive a.c. circuit the current and voltage are again 90° out of phase, but here the voltage lags behind the current—or, to put it another way, the current leads the voltage.

The reason why is easy to see. You learned in your study of time constant that, when the circuit is closed, the maximum current flows at once and the voltage across the condenser is zero. As current continues to flow, the condenser voltage builds up and, at the same time, the current decreases. When the condenser is fully charged the voltage across it is a maximum the current is zero.

When a.c. is applied, this same action occurs during one quarter of the oycle. Then, as the voltage decreases, electrons begin to flow off the condenser plates. By the time the voltage across the condenser reaches zero again, the condenser is completely discharged, and again a maximum current flow is taking place, this time in the opposite direction from the original flow.

Thus the circuit current and the condenser voltage are 90° out of phase in a capacitive circuit fed by a.c.; the circuit current is always a maximum when the voltage across the condenser is zero, and vice versa. Since the current is maximum first, the current leads the voltage.

Vector Diagrams. You remember that we had to use voltage vectors to combine the voltages across a coil and a lamp, because the two voltages were out of phase. Since the voltages across the condenser and resistor in Fig. 24Bare also out of phase, we must also use voltage vectors and a vector diagram to combine these voltages.

First, we can see that the same alternating current flows through both the condenser and the lamp as these parts are in series, so we will use the current as a reference value. We start by drawing our current reference vector, as shown in Fig. 25A. Remember—in a vector diagram the *length* of any vector is proportional to



FIG. 25. The starting or reference position is shown at A. Each complete revolution of a vector represents one a.c. cycle. By general agreement among radio and electrical men, vectors are always assumed to be rotating counterclockwise, opposite to the direction in which the hands of a clock move. Furthermore, the starting or reference position for all vectors is the vector position shown at A, which is a line going to the right horizontally from the center (O) of the vector diagram. In a series circuit, we almost always use current for our reference vector. To make the vector A represent the circuit current, we make the length of the vector proportional to the effective current value which would be indicated by an a.c. ammeter. We then put on the arrow head, and label it with the capital letter I to indicate that it represents current. As the resistor voltage drop is in phase with the current, its position will be the same as the reference line, so we draw it right on top of the current line, as at B.

the amount of current or voltage the vector represents. Further, the position of any current or voltage vector with respect to the circuit current reference vector we have just drawn shows how much the current or voltage is out of phase with the circuit current.

Since the resistor voltage drop is in phase with the circuit current, we draw the resistance voltage vector $V_{\rm R}$ right over the current vector, as shown in Fig. 25B.

We must now draw the vector for the condenser voltage drop. We just learned that voltage lags the current in a condenser by 90°, so we will have to draw the voltage vector V_c 90° behind our reference current vector, or straight down from the point of origin (O), as shown in Fig. 26A. Then we complete the rectangle, as shown in Fig. 26B, and draw a diagonal line from the point O to the opposite corner of our rectangle. This diagonal, vector E, then represents the source voltage. Since it is behind the circuit current vector I, we know that the source voltage lags behind the circuit current in a capacitive circuit.

If we draw $V_{\rm R}$ and $V_{\rm C}$ to scale so that every inch along them represents a certain number of volts, the length of line *E* will be equal to the source voltage figured on the same scale. Thus, Kirchhoff's Voltage Law applies to our capacitive a.c. circuit when we use a vector diagram to take phase into account.

Impedance. You have learned that resistors have resistance, while coils and condensers have reactance. Now reactance and resistance are somewhat alike, in that both act to oppose the flow of current. In other words, current flow through either of them produces a voltage drop equal to the product of the current multiplied by the resistance or reactance. However, as you know, the voltage drops produced differ in phase from one another.

Thus, the voltage drop produced by current flow through resistance is in phase with the current. The drop produced by current through an inductive reactance leads the current by 90° . The drop produced by current through a capacitive reactance lags the current by 90° .

You have already learned how to represent these phase differences in voltage drops by vector diagrams. In each diagram, a voltage vector represents the product of the circuit current multiplied by the resistance or reactance of the part concerned. Thus, $V_{\rm R}$ really equals $I \times R$, while $V_{\rm C}$ equals $I \times X_{\rm C}$ and $V_{\rm L}$ equals $I \times X_{\rm L}$. Now, since the current is the same in each of these voltage drops if the parts are in series (because the same currentflows through each part of a series circuit), we might just as well ignore it, and draw our vector diagram to represent our resistance or reactances only.

How do we do this? First, we draw our resistance vector R horizontal, making its length in inches proportional to its value in ohms. Then we draw our capacitive reactance vector X_c straight down, also making its length in inches proportional, on the same scale, to the ohmic value of the capacitive reactance. If we have a coil in the circuit instead of the condenser, we draw its inductive reactance vector





 $X_{\rm L}$ straight up, again making the vector length proportional to the ohmic value of the reactance.

Figs. 27A and B show how these vectors look when drawn this way. Notice —in each diagram, the rectangle has been completed and a diagonal vector Z drawn. What does this vector represent?

You know that, in the similar diagrams drawn for voltages, this diagonal vector represented the source voltage—the vectorial sum of all the voltage drops in the circuit. In these resistance-reactance diagrams, the diagonal vector similarly represents the vectorial sum of the reactance and the resistance in the circuit. It is called the "impedance" of the circuit; its symbol is "Z"; it is measured in ohms.

The impedance of a circuit represents the combined opposition of all the parts of the circuit to the flow of a.c. We can't add resistance to reactance directly, we must follow the same rules we do for their voltage drops. However, when we have the impedance, the product of the ohmic value of the impedance multiplied by the value of the circuit current in amperes equals the source voltage in volts.

Since this is true, we substitute impedance for resistance when we apply Ohm's Law to a.c. circuits. The three ways of writing the law then are:

OHM'S LAW FOR A. C.

$$I = \frac{E}{Z}$$

$$Z = \frac{E}{I}$$

$$E = IZ$$

As you will notice, they are exactly like the Ohm's Law forms for d.c., except that impedance is substituted for resistance.

Thus, if we know them, we can use impedance and the source voltage to figure the circuit current, and use this current with the resistance and reactance values to find the voltage drops across the parts. Or, if we already know the voltages across each part, we can add them vectorially to find the source voltage, as we did in Fig. 26. Either method will give us the right answer, but one will usually be more convenient than the other, as we shall see later.

CONDENSER LOSSES

Losses in the dielectric of a condenser keep it from being 100% efficient. You have already learned that there's a certain amount of leakage through the dielectric. This leakage provides a discharge path which will allow the charge to "leak off" and prevents the condenser from keeping its





charge forever. The better the condenser is, the lower leakage loss it has.

You can get some idea of how good a fairly large paper condenser is by connecting it to a 200- to 400-volt d.c. source long enough to charge it, breaking the connection, waiting two or three minutes, then shorting the condenser terminals with a screwdriver. If you get a strong spark, you know the condenser has low leakage and is therefore of good quality. (This test is useful only for paper condensers larger than .25 mfd.)

Another dielectric loss is caused by the fact that the dielectric absorbs some energy from the charging operation which it does not give back at once upon discharge. The reason is that when a condenser is charged, then discharged suddenly, all the bound electrons do not return at once to their normal orbits. If you wait a moment or two, then short the condenser again, there will be another (smaller) discharge. This effect is called "absorption." In an a.c. circuit, absorption is given the technical name of "dielectric hysteresis" (*di-eh-LECK-trick HISS-ter-E-sis*).

Both leakage loss and hysteresis loss can be lumped together and called the dielectric loss of the condenser. For simplicity, we can consider their <u>com-</u> bined effects as either a low resistance in *series* with a perfect condenser, or



FIG. 28. Dielectric losses are in series with the condenser, while leakage losses are across the condenser. One or the other may be ignored, or both can be lumped together and shown by either of these methods.

as a high resistance in parallel with a perfect condenser. Both ways of picturing losses are shown in Fig. 28.

The losses in a condenser go up as the frequency goes up, so condensers do not work as well at higher frequencies unless carefully made. Certain kinds of condensers cannot be used at high frequencies, particularly electrolytic condensers, due to the high amount of loss in these condensers.

Power Factor. Another way of expressing the losses in a condenser is in terms of "power factor." The power factor of a condenser shows the per cent of applied electrical power that is wasted by the resistance of the condenser. It is usually expressed in per cent. A perfect condenser would have a power factor of zero, while one with a high loss would have a fairly high power factor. As you can readily see, the lower the power factor of a condenser, the more efficient it is.

Test instruments are available that measure the power factor of condensers directly. Such instruments are used in radio service work, for they are very helpful in telling whether a condenser has such high resistance in series (high power factor) that it should be replaced. Usually such a test is necessary only on electrolytic condensers. You'll learn more about these instruments when you study service equipment in later lessons. For now, just remember power factor tells how much loss there is in a condenser.

Voltage Division with R, L and C

We have now reached the point where we can combine what we have learned about the action of resistors, coils and condensers into some of the basic circuits which you will meet time after time in radio receivers and other radio equipment. This section is a preview of circuits you will study again in later lessons. Several readings now, and later when you encounter similar circuits are recommended. ▶ In radio circuits we may have d.c. alone; we may have d.c. mixed with an alternating current; we may have d.c. with many different a.c. frequencies, or we may have different a.c. frequencies together. Regardless of what we have, we must have a means of combining and separating the different a.c. frequencies, as well as combining any or all frequencies with d.c., in order to obtain the radio actions we shall study later. In addition to making the proper combination and separation, we must consider the necessity of supplying the right voltage for the particular reaction we want to occur.

▶ Right now, we are going to start off with some very simple circuits. However, complex radio circuits can be



FIG. 29. When the source voltage is exactly correct for the load, they can be connected together as at A. However, a series part may be needed as at B, to lower the voltage, separate d.c. from a.c., or to separate different a.c. frequencies.

"boiled down" into simple circuits so that the actions in a radio receiver are easy to understand. We will leave the subject of complete radio circuits for later lessons, where you can pick them up one by one, analyze them into their basic elements, and thus reach a complete understanding of just what goes on in a radio. This thorough understanding of the actions which occur in a set will lead you directly into an understanding of what might go wrong, how it can go wrong, and what is necessary to clear up trouble when it arises.

▶ Suppose we start with a source and "load" connected as shown in Fig. 29A. (We call the device to which the source is connected a "load" because it uses the energy from the source.) For us to be able to connect a source of voltage directly to any radio part like this, the source must deliver voltage of exactly the right amount and right frequency for that radio part.

If the source furnishes too much voltage or voltages with the wrong frequency, this condition is often corrected by using the proper part in series with the load, as shown in Fig. 29B. The exact size and kind of series part to use depends on the correction needed.

SERIES RESISTANCE

Let us assume that our load is a resistor $R_{\rm L}$, as shown in Fig. 30, and that the voltage source supplies more voltage than we want to apply to $R_{\rm L}$. Now, suppose we put resistor R in series with $R_{\rm L}$. As you have already learned, the source voltage will divide, part appearing across R, the rest across $R_{\rm L}$ This same division will occur whether the source supplies a.c. or d.c., and the frequency of the source makes no difference. Thus, the effect of resistor R is to reduce the amount of voltage available for the load $R_{\rm L}$. The amount of reduction depends on the relative resistance of R and $R_{\rm L}$. Therefore, by adjusting the value of resistor R, we can adjust the voltage across $R_{\rm L}$ to the desired value. However, resistor R will affect only the amount of voltage. It has no effect on the fre-



FIG. 30. The series resistor can reduce the voltage to the right amount for the load but cannot separate frequencies.

quency of the source voltage, and cannot separate frequencies, so the source must deliver the correct frequency.

SERIES INDUCTANCE

Now, suppose we use an inductance L instead of resistor R, as shown in Fig. 31. We have again made up a voltage-dividing circuit, but one which acts in an entirely different manner.

▶ Suppose the source produces d.c. The normal resistance of the coil is relatively low, so there will be very little d.c. voltage drop in coil L. Nearly all the d.c. voltage will appear across $R_{\rm L}$. As far as d.c. is concerned, the coil L has practically no effect in this circuit.

▶ Now suppose the source furnishes a.c. of both low and high frequencies. The coil has a low reactance at low frequencies, so very little low-frequency voltage is dropped across it. But the reactance of coil L is very much greater at high frequencies, so most of the high-frequency voltage supplied by the source is dropped across the coil, and only a little appears across $R_{\rm L}$. Thus, the series coil acts as a frequency separator. It passes d.c. and low-frequency a.c., but tends to exclude high frequencies from the load. The larger the reactance, compared to the value of $R_{\rm L}$, the greater the tendency to exclude high frequencies

As low frequencies find less opposition in the coil, this circuit might be called a low-pass filter, because it passes low frequencies but filters out the higher frequencies.

▶ Again, as in Fig. 30, the division of voltage between the load and the series coil depends on the relative op-



FIG. 31. The coil has greater reactance as frequency is increased, so higher-frequency voltages are divided so most of the drop is across the coil. The coil thus tends to separate frequencies, excluding the higher ones.

position each offers to current flow. If the ohmic value of the coil reactance is high compared with the ohmic value of load resistance, more voltage will be dropped across the coil than across the resistor. And remember, the reactance of the coil depends upon both its electrical size and the frequency of the source. It is quite possible for the reactance of the coil to be much smaller than the resistance of the resistor at low frequencies and much higher at high frequencies. The coil reactance may be large because of high inductance, high source frequency, or both.

SERIES CAPACITY

Now let's change the circuit to that shown in Fig. 32, with a condenser C as the series part.

We know that if the source is a d.c. voltage, there will be an initial charg-



FIG. 32. The condenser blocks d.c. and, as its reactance decreases as the frequency is increased, it passes higher frequencies with less opposition, exactly opposite to the action of a coil.

ing current, the duration of which depends on the time constant of C and $R_{\rm L}$. However, once the condenser is charged, there will be no further d.c. flow through the circuit (except for a tiny leakage current, which is so small that we can normally ignore it). Therefore, we can say that the condenser *blocks* the d.c. flow because, as soon as the condenser is charged, there is no current in the circuit and no d.c. voltage across $R_{\rm L}$. All the voltage supplied by the source appears across condenser C.

▶ We have learned that a.c. does flow through a condenser, however. Further, we know that the higher the frequency, the *lower* the condenser opposition to a.c. flow.

Thus, the action of a series condenser is opposite to the action of a series coil. For low-frequency a.c., there is a large voltage drop across the condenser, leaving less voltage for the load resistance. At higher frequencies, less and less voltage is dropped across the condenser and more of the source voltage appears across the load resistance.

Again, the relative opposition of the condenser and the load to current flow determines the voltage division between them. The larger the capacity, or the higher the source frequency, the lower the condenser reactance—and the greater the voltage applied to the load.

Fig. 32 is one of the most common radio circuits you will meet later on. Condenser C is sometimes called a blocking condenser, because it prevents d.c. from flowing. As it will pass a.c. to the load, it is also called a coupling condenser, because it "couples" an a.c. voltage from one circuit to another. The name used depends upon which action is more important in the circuit.

A Typical Circuit. Fig. 33 is a typical radio circuit where a condenser is used both as a blocking and a coupling condenser. Battery B causes a d.c. flow through tube VT_1 and through resistor R_1 . Condenser C prevents direct current flow from battery B through resistor R_2 . However, a.c. signals in the circuit of tube VT_1 are passed on to R_2 through condenser C, where they operate tube VT_2 .

Of course, we don't expect you at this point to understand all about how this circuit works. There are a few practical facts you can see, though. If condenser C becomes "open" (disconnected by a break where the lead is fastened to the foil), no signal energy will be passed on to the next tube, because the circuit is broken so the condenser is effectively not there. This may make the receiver dead.

On the other hand, if condenser C becomes leaky or short circuited, d.c. will flow through this condenser and through resistor R_2 , where it is not wanted. As you will learn in a later lesson, this will cause distortion.

Thus, a properly operating conden-

ser C blocks d.c. and couples the a.c. signal to the next tube. A defective condenser C may prevent the passage of signal energy, or may permit the passage of d.c.

Summary. Fig. 34 shows in chart form the action of a resistor, coil and condenser individually. You will find this chart helpful in summarizing the information given to you so far.

▶ Now, having learned how these parts will act when in series with the load, let us consider a few more basic circuits using condensers.

BY-PASSING AND FILTERING

Suppose we go back to our resistive circuit shown in Fig. 30 for a minute. You will recall that the voltage division depends on the relative resistance values, but the same division occurs regardless of frequency. This circuit cannot separate d.c. from a.c., nor can it separate alternating currents of different frequencies.

▶ There are plenty of cases where we want d.c. in the load without a.c., or



FIG. 33. A typical resistance-capacitance circuit utilizing the basic circuit actions of Fig. 32.

want low-frequency a.c. and no high frequencies. We can add a condenser in parallel with the load, as in Fig. 35, and will again have a frequency-separating circuit---one of the most useful in radio.

As you know, there is no d.c. path through the condenser. If we apply a

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CONNECT R, L, OR C TO TERMINALS A AND B					
APPEARANCE	SYMBOL	DOES IT CONTROL D.C. CURRENT ?	DOES IT CONTROL A.C. CURRENT ?	WHAT HAPPENS WHEN THE ELECTRICAL SIZE IS INCREASED ?	WHAT HAPPENS WHEN FREQUENCY OF A. C. IS INGREASED ?
RESIS -TOR		YES	YES	OPPOSITION INCREASES	NO CHANGE IN OPPOSITION
COIL	we	NO (CURRENT PASSES)	YES	OPPOSITION INCREASES	OPPOSITION INCREASES
CONDENSER	-11-	NO (NO CURRENT PASSES)	YES	OPPOSITION DECREASES	OPPOSITION DECREASES

FIG. 34. This chart summarizes the actions of a resistor, coil or condenser when used in the simple circuit shown above. (Because a condenser completely blocks the flow of d.c., some engineers say that it does control d.c. to this extent.)

d.c. source, condenser C charges through resistor R, after which there is no further direct current flow in the condenser section of the circuit. Of course, the load R_L provides a d.c. path, so a direct current flows through resistors R and R_L , and the condenser might as well not be there, so far as d.c. is concerned.

When low-frequency a.c. is applied to this circuit, the capacitive reactance is high, and if it is much higher than the resistance of load R_L , very little a.c. flows through the condenser, most of it going through the load R_L .

As the frequency is increased, however, the reactance of the condenser decreases. Condenser C and resistor Rbegin to divide the source a.c. voltage, with a greater and greater drop occurring across resistor R as the frequency increases. Therefore, less and less a.c. voltage is available across C and the load $R_{\rm L}$. This circuit can be considered a low-pass filter, because low-frequency currents flow through $R_{\rm L}$, while higher frequencies do not.

Notice-this action is exactly opposite to that of the circuit in Fig. 32. The circuit in Fig. 32 permitted only higher frequencies to pass, while that in Fig. 35 permits only lower frequencies to pass. Hence, Fig. 35 is acting much like the coil circuit of Fig. 31. Sometimes we may use a coil as a filter, sometimes a condenser. By putting the coil or condenser in the proper place in the circuit, we can get either a low-pass or a high-pass action for a.c. The condenser has the advantage of blocking direct current flow when in series, and has no effect on d.c. when in parallel, so the condenser is often more desirable in circuits where d.c. is also present. You will usually find coils used more commonly in circuits containing only a.c. frequencies.

Notice how the parts work together to give the desired actions. Without the series R in Fig. 35, an entirely different action would occur. The condenser would then act just as it does alone and would serve only as an *additional load* on the source. It would not have any control over the $R_{\rm L}$ voltage unless the source contained enough impedance to replace the action of the series R. Of course, in practice, the source will have appreciable impedance so some a.c. drop will occur.

By-pass Action. The condenser connection shown in Fig. 35 is very frequently called a "by-pass" because it "by-passes" high-frequency currents so they do not flow through the resistor $R_{\rm L}$. This action is also called "filtering," because the effect is one of "filtering out" certain frequencies. Which name we use depends upon the exact action required in the circuit, as you will learn later. Usually, the term "by-passing" is used whenever we want to provide an easy path for a.c. and at the same time keep this a.c. out of another circuit.

Fig. 36A shows a typical by-passing circuit. Battery B causes d.c. to flow through the source, through R_L and through R_B . Condenser C has no effect on this action.

The source produces a.c., which we may want to keep out of battery B. This a.c. flows through R_L and condenser C, rather than through R_L , R_B and the battery, because the opposition to a.c. in the path through condenser C is very low, causing the a.c. voltage to divide between R_L and C. Since most of this voltage is across R_L there is



FIG. 35. Another important condenser circuit. Here, the condenser acts with resistor R to keep the higher frequencies from affecting the load RL.

little voltage across C, and as R_B and B are in parallel with C, they have the same low a.c. voltage across them.

The practical radio equivalent of this filter circuit is shown in Fig. 36B, where the tube VT acts as the source of the a.c. signal. Again, we include this circuit just to show you a practical use for one of our simple circuits. Much more information will be found in later lessons on this subject.

From what you've already learned, however, you can see that if condenser C in Fig. 36 becomes leaky, it provides a path for d.c. from battery B through $R_{\rm B}$. As the leakage resistance becomes lower, there will be a greater drop across $R_{\rm B}$ and less voltage will be available for the tube and $R_{\rm L}$. In other



FIG. 36. This typical by-pass circuit uses the basic actions of Fig. 35.

words, the d.c. path becomes $R_{\rm B}$ -C, rather than $R_{\rm B}$ - $R_{\rm L}$ -VT. Thus, a leaky condenser can stop the operation of the circuit. On the other hand, if the condenser becomes disconnected (open), the a.c. has no by-passing path and must flow through the battery, where it can produce undesirable effects, as we shall study later.

COMBINING L AND C

So far, we have discussed using a coil or condenser with a resistor. What happens when we combine a coil and condenser?

Let us first consider the circuit shown in Fig. 37. We know that the \cdot coil tries to exclude higher frequencies from $R_{\rm L}$, because it offers a greater opposition to higher frequencies. We also know that condenser C has less reactance at higher frequencies, and so tends to "by-pass" the higher frequencies.

Thus, both elements of this combi-



FIG. 37. Using a coil and condenser together this way gives a low-pass filter, which "passes" low frequencies to the load but sharply cuts off higher frequencies.

nation try to exclude higher frequencies from the load, L by offering high reactance and C by offering low reactance to the high frequencies. The practical result of combining these two parts in this circuit is that the high frequencies are cut off quite abruptly, instead of dropping off gradually as they do when either the coil or the condenser is used alone. This is a lowpass circuit, since only low frequencies reach the load.

By interchanging the two parts as shown in Fig. 38, we produce the opposite condition. Condenser C offers greater opposition to low frequencies, and coil L acts as a "by-pass" element. Therefore, the circuit in Fig. 38 cuts off low frequencies, and is thus a highpass circuit.

The actual frequency where this "cut-off" action occurs depends on the

values chosen for L and C in both circuits. When you take up filtering later on, you will find that coil and condenser combinations of these types are quite commonly used where it is necessary to separate high and low frequencies abruptly.

Looking Ahead. In radio, we will find circuits which contain resistance only. Circuits of this type will act the same regardless of the frequency. In other circuits, we introduce a coil or condenser to discriminate against either low or high frequencies. Then there are other combinations of coils, condensers and resistors which pass some frequencies but reject those higher or lower. Another group will reject a band of frequencies but allows higher and lower frequencies to pass.





Exactly when and how we do this will be the subject of many of your future lessons.

▶ In your next lesson you will take up the interesting and important subject of vacuum tubes and how they operate. You will then be ready to put together resistors, coils, condensers and tubes to form actual radio circuits.

THE N. R. I. COURSE PREPARES YOU TO BECOME A RADIOTRICIAN & TELETRICIAN (REGISTERED U.S. PATENT DIFIED (REGISTERED U.S. PATENT DIFIED

Lesson Questions

Be sure to number your Answer Sheet 7FR-3.

Place your Student Number on every Answer Sheet.

Send in your set of answers for this lesson immediately after you finish them, as instructed in the Study Schedule. This will give you the greatest possible benefit from our speedy personal grading service.

- 1. Name the four factors that determine the capacity of a condenser.
- 2. What is meant by the "working voltage" of a condenser?
- 3. Is it necessary to consider polarity when connecting electrolytic condensers?
- 4. How would you connect two condensers together to get an increased capacity?
- 5. When the frequency is decreased, does the capacitive reactance: 1, increase; 2, decrease; or 3, remain the same?
- 6. When the capacity is increased, does the capacitive reactance: 1, increase;
 2, decrease; or 3, remain the same?
- 7. When a resistor is connected in series with a condenser, does the charging time: 1, increase; 2, decrease; or 3, remain the same?
- 8. Suppose you want to exclude d.c. from a load. What part would you use in series with the load to do this?
- 9. Suppose a condenser is in series with a resistive load, as shown in Fig. 32. Will a larger-capacity condenser permit *more* or *less* a.c. voltage to be applied to the load?
- 10. Suppose distortion occurs due to a d.c. voltage across resistor R_2 of Fig. 33. Choose the two following conditions of condenser C which could cause this: 1, leaky; 2, open; 3, short-circuited; 4, normal condition.

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SUCCESS AND HAPPINESS

I would like to have you feel, as you read these short personal messages, that you are seated right alongside my desk. Years of experience with thousands of ambitious men proved to me that a word of advice or cheer can go a long way toward speeding your progress. As I see it, my responsibility goes farther than just giving you the very best training in radio—my duty is to help you get the very most out of *life*—to attain *real happiness*.

You, in common with all other N.R.I. men, desire success. You think that success will bring happiness, but this is not necessarily true. I believe that a man must train himself for happiness, just as he must train himself for success! Many a successful man of today is not happy, just because he did not realize this important truth.

The first thing you must understand is this: Happiness comes from within! There is no guarantee that material things—money, success, friends and possessions—will make you happy, for happiness is a state of mind. You must learn to be happy within yourself.

In these one-minute chats, then, I am going to teach you how to get the most happiness out of the success which is in store for you.

J. E. SMITH.





STUDY SCHEDULE NO. 8

For each study step, read the assigned pages first at your usual speed. Reread slowly one or more times. Finish with one quick reading to fix the important facts firmly in your mind, then answer the Lesson Questions for that step. Study each other step in this same way.

- □ 1. The Tube as an Electronic Switch......Pages 1-5 This section gives the ways in which electron emission can be produced, then describes thermionic emission in full. After learning this, you are shown how a diode tube can be used as a one-way device. This is the basic action of rectifiers and detectors—two important radio circuits. Answer Lesson Questions 1, 2, 3 and 4.
- 2. Basic Tube Construction......Pages 6-10 Here are the important facts about the kinds of materials used in making tube filaments, cathodes and plates. This is important information, because the materials used determine the conditions of tube operation. Answer Lesson Questions 5 and 6.

□ 4. The Triode Tube.....Pages 15-25 This section should be read and studied over and over. The triode is the basis for all multi-element tubes and is the basic amplifier tube. Be sure you understand exactly how the signal on the grid is used as the "pattern" for the tube to reproduce another and larger voltage within its plate circuit. Answer Lesson Questions 7 and 8.

- 5. Multi-Element Tubes......Pages 25-28 A brief introduction to tubes having many elements. You will meet these tubes in later lessons. Answer Lesson Question 9.
- ☐ 6. Special Purpose Tubes Photoelectric cells, gas-filled tubes and cold-cathode types are explained. Pages 29-30
- **8.** Mail Your Answers for this Lesson to N. R. I. for Grading.
- **9.** Start Studying the Next Lesson.

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World Radio History

HOW RADIO AND ELECTRONIC TUBES WORK

The Tube as an Electronic Switch

VACUUM TUBES are the heart of modern electronics. They are used not only in broadcasting and receiving radio entertainment, but also in commercial radio (aircraft, police, shipboard, and radio-telegraph systems), telephone repeater systems, diathermy equipment, and many other applications. In fact, tubes are used in devices which can see, hear, talk, feel, taste, smell, count, sort objects, keep time, control machinery, detect intruders, and cure diseases! And they range in size from tiny hearing-aid tubes scarcely larger than a thumbnail up to five-foot tall transmitter giants.

Let us now learn what kinds of tubes there are, how they operate, and what they can be made to do. These facts will apply to all tubes, whether they are used in broadcast, f.m., or television receivers or in transmitters. We must also learn something about tube weaknesses, since tubes are responsible for more radio breakdowns than any other part.

▶ When we have finished our study of tubes, we will be ready to put coils and condensers together in practical circuits, then to combine them with resistors and tubes to form actual radio circuits.

Electron Emission. In ordinary electrical circuits, electrons stay within the circuit wiring, and flow only over complete paths. However, in a tube, electrons are forced out of the surface of a metal cathode, and are made to flow across a space. Let's see how electrons are made to do these things.

▶ Electrons normally stay within a conductor, because an atomic force

prevents even free electrons from escaping through the conductor surface. An ordinary current flow through a conductor is a result of an *exchange* of electrons from atom to atom. There is no *loss* of electrons, because others replace those which pass on. However, if an electron can gain enough energy, it can separate itself from the atoms of the conductor and become a "free agent" in space.

There are four ways in which electrons can gain enough energy to escape into space from a metal or metallic compound. These are: 1, they can be evaporated or driven out by applying heat; 2, they can be driven out by bombardment with very small high speed particles, such as other electrons; 3, they can be driven out of some materials by the energy in light rays (which are electromagnetic waves); and 4, they can be jerked out by a very high positive potential.

All four of these methods are used in various types of electronic tubes to provide the free electrons on which all tubes depend for their operation. However, the first method (applying heat) is by far the most common, so we shall first deal with the *thermionic* tube. (Thermionic is pronounced THERM-I-ON-IK: thermo means heat, ionic refers to electrons.)

THERMIONIC EMISSION

Suppose we start with a tube consisting of a filament, made of thin resistance wire, enclosed in a glass or metal bulb from which all the air has been pumped. If we connect a battery to this filament (Fig. 1), the current flowing through the filament will heat it, and, when the filament has become hot enough, electrons will be emitted from it. (A vacuum is necessary in the bulb to prevent the filament from burning up at the temperature it will reach, and to remove the large air molecules which would interfere with electron emission.)

Electrons are emitted from the filament because the heat produced gives them enough energy to escape from the surface of the wire. The velocities with which they escape depend upon the amount of energy they get from the heat, and this starting speed determines how far they will go. Usually, they do not have velocities sufficient to take them very far from the filament, so a number of electrons soon are "hanging around" the filament. As this "cloud" of electrons be-



FIG. 1. When the filament is heated, electrons are emitted. Notice the symbols at B for the tube filament and for the battery.

comes thicker, they begin to repel each other, tending to force those closest to the filament back to the wire.

You learned in earlier lessons that removing electrons from an electrically neutral substance makes that substance positive. That is what happens to our wire filament—it becomes more and more positive as electrons are driven off from it by heat. This positive potential tends to attract electrons from the cloud back to the filament (unlike charges attract). Some of them do return, while others are prevented from doing so by the repelling action of the electrons freshly emitted from the filament.

Eventually a state of equilibrium will be reached, with a practically constant number of electrons in the cloud about the filament, and with the number of electrons leaving the filament approximately the same as the number returning to it. The electron cloud in the space around the filament is called the space charge.

▶ It is important to realize that the current from the battery in Fig. 1 just heats the filament. The electrons emitted by the filament do not come from this battery current—they come from the atoms of the filament wire itself. We could heat the filament by a gas flame, or any other heating agent, and get the same electron emission for the same amount of heat. An electric current is generally used to heat the filament because this is the most convenjent way to do it.

▶ Also—we show batteries only because they are a convenient way to indicate a *d.c. voltage* supply. This d.c. voltage might come from batteries, or from a power pack which converts a.c. to d.c., or from any other d.c. source. Don't think we are covering only battery type tubes—you are studying basic actions applying to all tubes. 1/

THE PLATE

Now that we have a filament surrounded by a cloud of electrons, let us place a metal plate within the tube and bring a wire from this element to the outside of the tube. Further, let us connect a meter M to this wire, then connect another battery between the meter and the tube filament. This will give us the circuit shown pictorially in Fig. 2A, and schematically in Fig. 2B. For identification, we call the filament-heating battery an A battery, and the new battery the B battery, as marked in Fig. 2B. Suppose we first connect the negative terminal of the B battery to the new element (which we shall call a plate), and the positive terminal of this battery to the filament, as shown in Fig. 2. (The circuit is completed from the negative B terminal to the plate through the meter.) We will



FIG. 2. Here the "plate" has been added. The meter indicates no current flow because the B battery has the wrong polarity.

find that the meter pointer remains at zero, showing that there is no electron flow in this "plate" circuit. There are two reasons for this. First, there is no electrical connection within the tube between the plate and the filament, so there is no metallic path for current (electrons) to follow. Second, the plate and the filament assume the potentials of the B battery terminals to which they are connected: the plate therefore becomes negative with respect to the filament. The negative plate will then repel the negative electrons emitted from the filament, so that they will not go to the plate and there will be no conduction through space in the tube.

Now, let us turn the *B* battery around (Fig. 3) so that the plate becomes *positive* with respect to the filament. At once, the meter will indicate that a current is flowing—because now the negatively charged electrons are attracted by the *positive* plate (unlike charges attract) and are pumped through the circuit by the *B* battery so that they flow through the meter M, through the B battery, and back to the filament.

In other words, free electrons in space obey the laws of charges-they are repelled by a negative plate, and are attracted by a positive plate. The number of electrons attracted depends on the voltage of battery B and on the distance between the plate and the filament. Increasing the voltage of battery B (thus making the plate more positive), or decreasing the distance from the plate to the filament, or both, increases the ability of the plate to attract electrons. Conversely, decreasing the voltage or increasing the distance decreases the attracting ability of the plate. (Of course, the distance between the plate and the filament is



FIG. 3. The proper B supply polarity causes an electron movement from filament to plate, through the meter and the B battery back to the filament. (There must be a complete path back to the filament for there to be an electron flow. The B battery can "take in" at its positive terminal only the same number of electrons as it "lets out" of its negative terminal. Hence, free electrons are given back to the filament in exchange for those emitted.)

always fixed in any particular tube, so in a practical case, we generally consider that the voltage applied to the plate determines the ability of the plate to attract electrons.)

What has happened to our space charge? Assuming average conditions, the electrons which go to the plate come from the space charge, and thus reduce the negative charge in the cloud around the filament. This, in turn, allows an equal number of electrons to leave the filament to make up for this deficit in the space charge. Thus, there is a transfer action in which electrons leave the filament and go to the space charge, while other electrons leave the space charge and go to the plate. For most purposes, we can ignore this transfer and assume that electrons go directly from the filament to the plate, when the plate is made positive-with respect to the filament by a B battery.

This electron movement continues through the meter and the battery so that an equal number of electrons are restored to the filament to make up for those emitted. Hence, there is a "complete circuit" effect when the plate is positive with respect to the filament.

▶ All vacuum tubes have at least two electrodes, a <u>cathode and an anode</u>. The cathode is the electron emitter, and the anode (or plate) is the electron-attracting element. If these are the only electrodes in the tube, it is called a diode tube. "Diode" is pronounced DI-OAD (rhymes with "load"): "di" means two, and "ode" refers to electrodes or elements—hence a diode is a two-element tube.

RECTIFICATION

So far, we have described how it is possible for electrons to flow from the cathode to the plate in a tube when the plate has the proper polarity. Notice —the tube is obviously a "one-way" device, in that electrons can move only from the cathode to the plate, and then only when the plate is positive with respect to the cathode. Electrons do not normally move from the plate to the cathode.

▶ This one-way action makes a diode tube an important device in a.c. systems. Suppose we substitute an a.c. supply for the *B* battery, giving us the eircuit shown in Fig. 4A. As you know,

an a.c. voltage reverses polarity periodically; so for one-half of the cycle, terminal 1 will be positive with respect to terminal 2, while for the other halfcycle, terminal 1 will be negative with respect to terminal 2. If our supply voltage is an a.c. sine wave, like that shown in Fig. 4B, we will find that plate current flows in pulses like those shown in Fig. 4C. In other words, plate current flows only for the portion of each voltage cycle when terminal 1 is positive with respect to terminal 2, and does not flow for the half of each voltage cycle when terminal 1 is nega-





tive with respect to terminal 2. In fact, we might consider the tube to be a kind of "electronic switch" that closes the tube circuit (and so permits current flow) only when the plate-tocathode voltage has the proper polarity.

Thus, plate current flows in only one direction, and so is a *direct current*. The current is *pulsating*, varying from zero to some peak value which depends on the supply voltage, but it never reverses in direction.

▶ This is an extremely important characteristic of a tube. When we use an a.c. voltage supply in the plate circuit, the plate current will be a pulsating d.c. current. If this d.c. current is made to flow through a resistor, it must produce a pulsating d.c. voltage drop across the resistor. Thus, the



tube *rectifies* an a.c. voltage—that is, changes it to a d.c. voltage.

Fig. 5A shows the same circuit given in Fig. 4A, except that the meter in the latter has been replaced by a "load." This load might be a resistor, or a coil, or a complex electrical circuit. The tube in Fig. 5A acts, as you have just learned, as a one-way device and permits electron flow only in the direction indicated by the arrows, (and then only when the supply



FIG. 5. The load voltage polarity is reversed by reversing the tube connections.

voltage polarity makes the plate positive). Hence, the voltage drop across the load will be a pulsating d.c. voltage which has the polarity shown. (By the use of filter circuits, which you will study later, it is possible to "smooth out" the variations in the current, so that the voltage across the load is practically a steady d.c. voltage like ' that delivered by a battery.) Ni ▶ Should we want the load voltage polarity to be reversed from that in Fig. 5A, we need only reverse the plate and cathode connections as shown in Fig. 5B. Electrons still can go only from cathode to plate, so this reversal of the tube connections changes their direction through the circuit.

Practical Uses. Radio finds a number of important uses for the oneway action and for the basic circuits shown in Fig. 5. For instance, we can take power from an a.c. electric line and change it into a d.c. voltage which we can use to run a radio. This has made it possible to eliminate batteries when we wish to operate the radio from a power line. When a diode tube is used for this purpose, it is called a rectifier, and the circuit in which it is used is known as the "power pack" of the radio receiver.

Diode tubes have many other uses in radio which need not concern us at the moment. You will meet them all in later lessons. The one-way action of a tube is of great help to servicemen. It is frequently important to know the polarity of voltage drops in tube circuits. Now that you know electrons can pass through the tube only from the cathode to the plate, you can determine the direction of electron flow in any tube circuit just by looking at the tube. Then you can trace around the circuit and determine the polarity of the voltage drop in any device from the rule you learned earlier:

The end of any coil, condenser, or resistor which electrons enter is the negative end of that device.

For practice, use this method to find the polarity of the voltage drop across the load in Figs. 5A and 5B. You should arrive at the polarities indicated in these figures.
Basic Tube Construction

All tubes have a cathode and a plate, and all thermionic types have a filament. Since these elements are of such basic importance, let us take a few moments to learn just how they are made. This will teach you some important facts about the uses to which tubes can be put, and also the reasons for certain weaknesses which they have.

TUBE FILAMENTS AND CATHODES

The earliest tube filaments were made from pure metals. It was necessary to make these pure metals extremely hot before electrons could be forced out of the surface of the filament. Hence, it was necessary to find metals which would emit sufficient electrons for electronic tube purposes without being melted by the intense heat.

Pure Metals. The pure metal most commonly used for tube filaments is tungsten. This metal is capable of withstanding very high temperatures, but is a relatively inefficient electron emitter because a considerable amount of heat is necessary to get the desired degree of emission. This means the filament-heating A supply must furnish a great deal of power.

Today, pure metal filaments (used as electron emitters) of this type are found only in high power transmitter tubes, where the ability to withstand high temperatures and the freedom from gas troubles (to be discussed later) make such filaments desirable.

Thoriated Filaments. After the development of the tungsten filament, it was discovered that certain impurities in the metal improved its electron emitting properties. Tungsten is a hard, brittle metal, and when pure is difficult to draw into the form of wire. Engineers found that a little thorium mixed with the tungsten overcame the brittle properties, giving a more sturdy filament for light bulbs. Radio engineers tried these thoriated tungsten filaments and found them not only sturdier, but also much better as electron emitters than pure tungsten filaments.

The manufacturers subject a thoriated tungsten filament to a process called "activating" when the tube is constructed. In this process, the filament is first subjected momentarily to a high voltage overload, and then to a small voltage overload for a considerable period of time. The high flashing voltage at the beginning causes some of the thorium to flow from within the filament to the surface, and the baking process which follows causes this thorium to form a layer on the surface of the filament wire. This layer (which is about one molecule thick) has the property of increasing the electron emission tremendously. The temperature required to produce a desired number of electrons from a thoriated



FIG. 6. Typical filaments.

tungsten filament is far lower than the temperature required to produce the same emission from a pure tungsten filament.

At one time thoriated filaments were used in practically all tubes. Today, they are found mostly in medium power transmitting tubes, having been replaced in the lower-power class and in receiving tubes by oxide-coated filaments.

Oxide-Coated Filaments. The oxide-coated filament is made by coating a filament wire of nickel or a platinum alloy with the oxides of certain metals. Oxides of barium, strontium, and calcium have been used.

Oxide coatings can be used only in tubes where the plate voltage is below 500 volts, for higher voltages jerk away the coating. Also, they can be used only in tubes where there is little chance of gas trouble. However, they are entirely satisfactory for receiving tubes and low-power transmitting tubes, where both these conditions are met.

Oxide coatings are by far the most efficient electron emitters, requiring less heat than any other kind of surface for the desired degree of emission. This means that a low-power filament source can be used with an oxidecoated cathode. In fact, so little heat is needed that oxide-coated cathodes have been developed which are indirectly heated. These have proved very useful in reducing hum when tube filaments are operated from an a.c. source. (A.C. power will supply heat just as well as d.c. power, but hum is caused if the emission varies with the a.c. cycle changes. As you will learn later, a very heavy filament or an indirectly heated cathode is necessary to eliminate this hum voltage variation.)

▶ Now is a good time to clear up the meanings of the words "filament" and

"cathode." The *filament* is the proper name for the resistance wire supplying the heat. The electron emitter is called the cathode. When the filament does the emitting, it is also the cathode. Several different filaments, which are also cathodes, are shown in Fig. 6.

▶ In the indirectly heated cathode, the filament merely furnishes heat to the electron-emitting cathode, which is a separate element. The construction of several indirectly heated cathodes is shown in Fig. 7. The filament is threaded through holes in a ceramic insulator, which is placed inside a nickel-alloy sleeve or "thimble." This metal sleeve has the oxide coating



FIG. 7. A separately heated cathode emits electrons.

baked on it. (In some types, the filament wire is coated with the ceramic insulation and is then placed inside the sleeve.) The insulation prevents there being an electrical path between the cathode and the filament. Since the sole purpose of the filament in this construction is to supply heat, it is commonly called a heater. A tube using an indirectly heated cathode is frequently called a "heater type" tube, to distinguish it from the "filament type" which uses the filament as the cathode.

▶ Metals stretch when heated, so the filaments in all tubes are mounted on spring supports which take up the "stretch" of the filament wire, thus preventing it from sagging and touching other elements; yet the springs allow it to shrink back to normal when the tube is not heated.

Tube filaments are delicate; a severe jar or excessive voltage will cause them to break or burn out. The proper



FIG. 8. Typical tube structures.

filament voltage is so important that the tube type numbers indicate the amount to be used, as we shall see later in this lesson.

VACUUM TUBE PLATES

The tube plate is subjected to considerable heat, some radiated from the filament and some produced by the electrons bombarding the plate itself. When electrons strike the plate, they give considerable energy to it. If the plate becomes hot enough, it too will begin to emit electrons. In fact, the one-way characteristic of the tube is dependent upon whether the plate remains relatively cool. Nickel, molybdenum, carbon, and pure iron are materials that have proved suitable for use in constructing plates or anodes. The plate is generally given a dull black surface, for black surfaces radiate heat readily and therefore keep cooler than polished surfaces.

The plates of medium power tubes are often fitted with fins to improve their heat radiation. In high power tubes, water cooling or forced air draft is necessary to keep the plates at safe low temperatures.

▶ Even with the plate cool, the electrons striking it can knock other electrons loose from its surface. This effect, wherein speeding electrons knock loose other electrons from the surface of an element, is called <u>secondary</u> <u>emission</u>. If the plate or anode is the only positive element in the tube, this condition is not serious, as the secondary electrons will go back to the plate. In other tubes, special means are taken to eliminate the undesirable effects of this secondary emission. We shall learn what these are a little further on in this lesson.

▶ The plate shape depends somewhat on the shape of the cathode and on the structure of the other electrodes in the tube. In early tubes, it was just a flat plate off to one side. Now, it always completely surrounds the emitting sections of the cathode. If the cathode is a vertical wire or is indirectly heated, the plate will be a cylinder which surrounds the cathode, as shown in Figure 8A.

Where the filament is constructed like an inverted V or W (as it is in Fig. 6B or 6C), the plate will generally be an oblong or oval structure like that in Fig. 8B.

GAS EVACUATION

The normal tube has a very high degree of vacuum. If air is permitted in the tube, the filament will oxidize (burn up) when heated.

▶ An even more important effect of atmospheric gases within a tube is the fact that the speeding electrons will strike the gas molecules, knocking other electrons out of them. This changes the molecules into large, heavy, positive, gas ions. The cathode is negative, so the positive ions will travel to the cathode and bombard

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it. Oxide-coated cathodes are easily ruined by such bombardment, for the oxide layer is knocked off completely. A thoriated filament is similarly destroyed, although not as easily. A pure tungsten filament is the only type capable of withstanding much gas ion bombardment.

▶ To eliminate the effects of gas as much as possible, vacuum pumps are used to draw out air and gases. During the evacuation process, the elements within the tube are heated by induction heating apparatus, which uses large coils to induce voltages in the elements. The resulting heat tends to drive out much of the gas. We ordinarily think of metals as being absolutely solid, but they have pores which will hold gas molecules until they are driven out by such heating.

In addition to the heating and evacuation, most tubes employ "getters." A getter is a small cup containing chemicals. During the induction heating process, these chemicals vaporize and combine readily with the gas molecules, forming metal compounds which then are deposited on the cool glass envelope of the tube. This is what gives the silvery appearance many tubes have near the base of the bulb. These compounds "hold on" to the gas molecules and do not readily release them in the tube space.

▶ Tubes having oxide-coated filaments cannot be heated to as high a temperature as can those having other filaments (the oxide coating will be boiled off if too much heat is applied). so the gas evacuation is not as good in these tubes. As a result, oxidecoated cathodes can be used only in tubes that are intended for relatively low power and low plate voltage. (As the plate voltage is increased, the electron velocity is speeded up and the heating effects are increased, so that there is more danger of electrons ionizing gas molecules present in the tube space and also of knocking out gas molecules from the metals.)

A tube with a pure tungsten filament, on the other hand, can be very highly evacuated, because tungsten is not affected by high temperatures. This is one reason why high power tubes, particularly transmitting tubes operating with plate voltages above 5000 volts, usually have pure tungsten filaments.

▶ While every effort is made to eliminate gases from within *vacuum* tubes, some special "gas-filled" tubes do have certain gases deliberately introduced in them. We will study these tubes, and the reason for putting gas in them, later in this lesson.

▶ Since the tube must be sealed completely, the wire leads going to the ele-



FIG. 9. A comparison of the construction of a glass envelope tube and a metal envelope tube. 9

ments must pass through a seal formed by a glass press or glass beads before emerging from the tube. Considerable research was necessary to find a metal to be used for these wire leads which had the same rate of expansion as the glass, so heating would not crack the seal. These wires are special alloys. The supports and connectors to the elements are welded to the wires within the tube.

Fig. 9 shows the method of sealing both a glass-envelope tube and a metal-envelope tube. A metal tube differs from a glass tube in having

a metal envelope instead of a glass envelope, so a metal tube and a glass tube of the same type are electrically identical inside their envelopes.

After the gas is evacuated, the tubing which goes to the vacuum pump is sealed off and cut. The leads emerging from the tube are then fastened to prongs in a base, so that the tube can be plugged into a socket. Circuit connections are made to the socket, so the tube elements will be connected to the tube circuit when the tube is plugged into the socket.

Basic Tube Characteristics

So far, you have learned how the diode tube can be used as a rectifier. You have also learned some facts about the constructional features of these tubes, which also apply to all other types of tubes.

Now, let us learn more about how tubes operate. For simplicity, we shall use the diode as an example at first, then take up other tube types when you have mastered the important fundamental facts of tube operation.

FILAMENT VOLTAGE LIMITS

Suppose we set up the circuit shown in Fig. 10 to see just what happens as the filament and plate voltages are varied. (We have shown an indirectly-heated cathode type tube in this circuit, but might equally well have used a filament-cathode tube; the following explanations apply to either type.)

Let us start out with a relatively low plate-to-cathode voltage from battery B. (Radiomen shorten this, calling it just "plate voltage." You know voltage exists between *two* points, so the names "plate voltage" or "grid voltage" refer to voltages between these points and some reference point, which is usually the cathode.)

Let us also set the variable resistor R in the filament circuit to a high value, so that very little current flows in the filament circuit. The tube filament and resistor R form a voltage divider. When the resistance of R is high, most of the voltage is across it and meter V shows very little voltage across the tube filament.

With very low filament voltage, the filament will not get hot enough to cause cathode emission, so the meter I will indicate zero plate current. Suppose we now reduce the resistance of rheostat R gradually. This lets more and more current flow through the filament circuit, and a greater proportion of the voltage appears across the tube filament. As a result, the filament becomes hotter, thus raising the temperature of the cathode. Soon meter I will indicate that a current is flowing in the plate circuit. This plate current means, of course, that electrons are being emitted by the cathode.

As we continue to decrease the value

of resistor R, the filament voltage continues to *increase* and, for a time, so does the plate current indicated by the meter I. However, we will eventually reach a point where the plate current no longer increases, even though the filament is made hotter.

Fig. 11 shows a graph of this action. As the filament voltage (E_t) is increased, the plate current (I_p) increases from A to B on the solid curve. At point B, the plate current ceases to increase much and the curve levels out toward C.

Now suppose we start the experiment over, this time with a higher voltage from battery B in the plate circuit. We will again find that the plate current follows the curve from A to B, but now it goes on up along the dotted line to point D, where it levels out toward E. The higher plate voltage results in a higher plate current (the distance from B to D) before



FIG. 10. A test circuit like this is used to determine the characteristics of diode tubes.

the curve levels off. Eventually, however, the leveling occurs.

Should we continue this experiment, increasing the value of the plate voltage each time, we would find that the events repeated themselves. Higher plate voltages would result in the curve rising higher before it leveled off. However, it would always level off at some point.



FIG. 11. How the plate current varies with different filament-heating potentials.

Temperature Saturation. We know that as the filament voltage is increased, the filament becomes hotter and hotter. This means more and more electrons are forced out of the cathode, so we might expect the plate current to continue to increase instead of leveling off. But, since the plate current finally becomes constant, evidently a point is reached at which only a constant number of the electrons forced out of the cathode go to the plate. Why don't all of them do so?

The answer lies in the space charge effect we have already studied. An electron newly emitted from the cathode is subjected to the attracting force of the plate (tending to pull it toward the plate) and also to the repelling effect of the electrons emitted just ahead of it (tending to push it back toward the cathode). When emission is low, all the emitted electrons are drawn to the plate. But as emission increases, the slower moving electrons begin to form the space charge. As the emission is further increased, electron repulsion by the space charge increases (because there are more electrons exerting a repelling force) while the attracting force of the plate remains constant. As a result, the net force pulling newly-emitted electrons to the plate decreases; this slows these electrons down, and they tend to bunch

up and join the electron cloud around the cathode.

Eventually, as emission increases, this electron cloud becomes so dense that we get the space charge transfer effect we described earlier. The plate pulls a constant number of electrons out of the space charge at any given instant; this same number of newlyemitted electrons can then enter the space charge, but any excess of newlyemitted electrons over this number is forced back to the cathode. The number of electrons the plate pulls out of the space charge depends upon the plate voltage and the distance from the plate to the space charge, but not on the emission.

Increasing the voltage on the plate will increase the current—because it will increase the number of electrons the plate can pull out of the space charge at any given instant. This merely results in a constant plate current of higher value, however, with the space charge still forcing excess electrons back to the cathode.

This means that for a particular plate voltage, there is a definite limit to the amount of current which will flow in the plate circuit of the tube. The only way you can get more current in a diode tube is to increase the plate voltage. Making the filament hotter will not increase the current greatly above the point where the curve begins to flatten out. The "bend" at point B on the solid curve (and at point D on the dotted curve) is known as the knee of the curve. The flat region of the curve beyond the knee is known as the saturation region. (A radio tube is saturated when the plate can attract no greater number of the emitted electrons.)

This condition, in which increasing the filament voltage will not cause a further plate current increase, is known as filament saturation or temperature saturation. Filament Voltage Ratings. If a filament is heated too high it will, of course, melt. Hence, we cannot continue to increase the filament voltage indefinitely. In fact, the tube filament is designed to have a resistance such that it will reach a desired temperature when a particular rated voltage is applied to it. (As you will learn in a moment, the rated filament voltage is different for various tube types.)

The temperature the filament will reach is made high enough so that, for any normal plate voltage, there will always be more electrons emitted from the cathode than can be attracted to the plate. This makes the plate current depend only on the plate voltage, not on the emission. In other words, tubes are usually run well into the filament saturation region, so we'll be sure to have all the electrons we need.

What filament voltage values are used? The first tubes had to use battery power supplies, so filaments were designed to operate from 6-volt storage batteries or from groups of dry cells. The first storage battery types had 5-volt filaments, and a series resistor was used with them to drop the voltage from a 6-volt battery. The resistor value was reduced as the battery voltage decreased with age—a scheme which permitted longer operation before the battery had to be recharged.

The early tungsten filament tubes required a high filament current, because the filament had to be made quite hot before sufficient emission was obtained. However, when the more efficient thoriated tungsten and oxide-coated filaments were developed, lower temperatures could be used and less current was required for filament supply.

► The development of a.c. power supplies freed tube filaments from the necessity of operating from battery voltage values. Step-up or step-down

transformers readily deliver any desired voltage. Today, we have a.c. tubes with filaments rated at 2.5 volts, 6.3 volts, 12 volts, 25 volts, 35 volts, 50 volts, and even 117 volts. (The last named type can be connected directly across a 110-volt power line.)

You may wonder how a.c. can be used on the filament without causing plate current variations as the filament voltage varies. The answer is that filaments and cathodes are used which hold heat for an appreciable period of time. Battery tubes will warm up almost instantly, but it takes 30 to 90 seconds for a.c. tubes to "get going." These tubes hold to a practically steady temperature during the small time space of an a.c. voltage variation, so there is practically no emission variation. We have to be careful with battery tubes, however, to see that a steady d.c. is applied to the filament. ▶ The numerous filament voltage values are the result of design requirements-some were developed for a.c.d.c. receivers; some for auto sets; some for other purposes and reasons. Modern portable battery-operated receivers were made possible by the development of an entirely new line of 1.4- and 3-volt tubes, using far more efficient oxide-coated filaments than earlier types. These tubes have such lowcurrent requirements that they can be run for a long time on small, lightweight batteries.

Transmitting tubes and other special purpose tubes have a similarly wide range of filament voltage ratings, so that a suitable tube can be found for each application.

Voltage Tolerances. Modern a. c. type tubes will operate over a fairly wide range about their rated filament voltages. For example, tubes with filaments rated at 6.3 volts will operate satisfactorily on voltages ranging from 5.8 volts to as high as 7 volts. With filament voltages below this

range, the emission may become too low for most uses, and with voltages above the range, the tube life may be shortened. (Practically all these tubes have oxide-coated cathodes from which the coating will boil off if the temperature gets too high. This will destroy a tube just as surely as a complete burn-out of the filament.) Naturally, if the voltage is too much higher than the rating, the filament will actually melt.

Modern 1.4-volt battery-type tube filaments are designed to operate from a single dry cell which has a rating of 1.5 volts, but which actually delivers 1.65 volts for a short time, then drops down to its 1.5-volt rating. The battery voltage gradually decreases to 1.2 volts; then the battery is supposed to be replaced. These tubes thus are intended to operate in the range from 1.2 to 1.65 volts. However, some circuits may cease functioning if the voltage drops below the 1.4-volt filament rating. A check of the filament voltage should be made in receivers using these tubes when they fail to work.

PLATE VOLTAGE LIMITS

Suppose we return to our circuit of Fig. 10 and this time apply the rated filament voltage so that the tube is operating in the temperature saturation region. Let us now start with zero plate voltage by moving tap 2 on plate battery B up to point 1. This means that none of the battery B will be in the circuit.

Under this condition, we will find that the meter I indicates zero plate current. Now, let us move the tap 2down battery B a step at a time. As we gradually increase the voltage difference between the plate and cathode, we will find that the plate current increases slowly at first, then more rapidly and steadily as the plate voltage is increased. Finally, the plate current increase begins to taper off. A plot of these results (Fig. 12) will give us a curve which is very similar to that shown in Fig. 11. This curve shows that for a fixed filament voltage, increasing the plate voltage above a certain value no longer gives proportional increases in plate current.

Voltage Saturation. As the plate voltage is increased, more and more



FIG. 12. How the plate current varies with the plate voltage.

electrons are drawn from the electron cloud. As we continue to increase the plate voltage, the space charge becomes less and less, thus losing its repelling effect on the electrons coming from the cathode. As a result, we will eventually reach the point where every electron the cathode can emit (at that temperature) is being drawn at once to the plate, and the space charge no longer exists. When this occurs, there is no way to increase the plate current further, except by making the filament hotter. We then have what is known as voltage saturation (increases in plate voltage do not produce proportional increases in plate current).

If we keep the filament temperature fixed, then voltage saturation places a limit on the plate current. However, with many tubes it is impossible to reach saturation without damage to the tube—because as the voltage is increased, the plate current becomes so high that the electrons bombarding the plate can heat it to the melting point. Another factor which limits the allowable plate voltage is the spacing between the plate and other elements, and the spacing between the leads coming through the glass press or seal at the <u>tube</u> base. If the plate voltage becomes too high, it is possible for an arc (or electric spark) to jump between the elements. Should this arc jump between the leads in the seal, the tube will be destroyed.

The maximum plate-to-cathode voltage is always specified for each type of tube. Then, various sets of "operating voltage" values are given, which are equal to or lower than this maximum.

This rating limits the a.c. voltage we can apply in the circuit shown in Fig. 5. It also limits the d.c. voltage we can apply, as we shall learn when we take up tubes requiring d.c. plate voltages. Incidentally, since the early tubes were intended to operate from batteries, and the standard B battery is a 221/2 or 45-volt battery, d.c. operating plate voltages were given in multiples of these values. Thus, 45, 671/2, 90, 135, and 180 volts were common operating voltages for many tubes in the early days of radio. With the advent of a.c. power supplies, which can furnish any desired d.c. voltage, tubes were free of the necessity of operating from battery voltage values. Today, the most common receiving tube plate voltage rating is 250 volts, while transmitting tube ratings range up to 10,000 volts. Tubes intended for both battery and a.c. power pack operation are still rated at plate voltages easily obtainable from batteries

PRACTICAL FACTS

From what you have learned, you can see that there are limits to the plate and filament voltages that can be used with any particular tube. Should a tube be needed for a particular application, it is important to choose one having the proper ratings. This need for tubes designed for special purposes has led to the development of over 500 receiving type tubes alone.

▶ Fortunately it is not necessary to learn the values of the characteristics of all these tubes. You need to know only what these characteristics mean and the limits they place on tube operation. Then, like all radiomen, you can refer to a tube characteristics chart and can look up the values for any tube in which you are interested. We will go into this more thoroughly later, after we have learned more about some of the other types of tubes.

The Triode Tube

The triode (TRY-OAD) or threeelement tube is perhaps the most important development in radio's history. The additional element used in the triode makes amplification possible. Also, this tube started the development of other multi-element tubes which have made possible circuits undreamed of in the early days of radio.

You learned in earlier lessons that a radio wave, after it is picked up by an antenna, reaches the receiver in the form of an a.c. voltage. This voltage is called a "signal" voltage. It is usually very small, and must be amplified many times before it can be used to operate a loudspeaker and give us sound. In all radios except the very simplest crystal sets, this amplification of the signal voltage is furnished by tubes. The signal is fed into a tube, and an amplified signal is developed across a load in the tube plate circuit.

Can we use a diode to amplify an a.c. signal? Going back to the diode circuit shown in Fig. 5, you can see that the load voltage could never exceed the a.c. source value. (Kirchoff's Voltage Law: The voltage rises equal the voltage drops.) In fact, since there is always some loss in the tube itself (it acts with the load as a voltage divider), you actually get less voltage across the load than is furnished by the source. Therefore a diode can-

not be used to amplify an a.c. signal, for the voltage developed across the load will always be smaller than the signal itself.

The triode, however, can be used to amplify a signal. With this tube, the signal is used to CONTROL the voltage furnished to a load by a separate power source, and even a very small signal can be made to control a considerable voltage. This action is obtained by applying the signal voltage between the cathode and an additional element within the tube.

HOW THE GRID WORKS

This new element is called a grid, and, as its name implies, is of open construction. This grid may be a spiral wire with large spaces between the turns, or it may be a wire mesh something like a window screen. Several typical grid constructions are shown in Fig. 13. Since the grid is an alternate "wire and space" affair, the grid symbol on a schematic diagram is usually like that shown in 13D, although some engineers use the 13E symbol.

The grid is positioned closer to the cathode than it is to the plate. (This spacing is not shown on schematic diagrams because the same symbol is used for all triodes, regardless of the actual spacing.) Before studying how amplification is obtained, let us see how the grid can control the plate current. We will apply different voltages to this grid and see what happens.

Zero Grid Voltage. Suppose we first connect the grid to the cathode as shown in Fig. 14. Connected directly by a wire this way, these two elements have no voltage between them.

If we apply a normal plate voltage from battery B between the plate and cathode, we will get an electron flow, and this new element will have very little to do with it. The electrons are pulled along by the positive plate potential and only those moving di-



FIG. 13. Different grid structures and the symbols used to represent the grid element.

rectly toward a grid wire will strike it and either be deflected or captured. Those electrons traveling toward the *spaces* between the grid wires will go right through these spaces and on to the plate.

Positive Grid Voltage. Suppose we now put a small battery in the grid circuit, between the grid and cathode, as shown in Fig. 15. You will recall that the attracting power of the plate on an electron depends both on the plate voltage and on the distance between the plate and the electron, and this attracting power is greater the closer together the plate and electron are. Exactly the same thing is true of the grid; it, too, has a greater attracting power on an electron the closer it is to the electron. Since the grid is so much closer to the source of electrons (the cathode) than the plate is, naturally the grid has far more effect on electrons emitted from the cathode than the plate has if we apply equal voltages to both the grid and plate. As a matter of fact, the difference in effect is so great that we can apply a far larger voltage to the plate than to the grid, and still the grid will have more influence over the emitted electrons. Thus, the grid in Fig. 15 will attract a great many electrons from the cathode-many more than the plate does.

If the grid were a solid plate, it would capture all of the electrons it attracts toward itself and there would be a large current in the grid-cathode circuit. However, the open grid construction prevents this; again, only those electrons which happen to strike the grid wires are captured. The others tend to go in straight lines at high speed right through the spaces between the grid wires. Once well past the grid, they come under the influence of the plate, which then collects them.

This means that when the grid is made slightly positive, it increases the number of electrons which go to the plate.

As the grid is made more and more positive, the plate current increases further. However, as the grid becomes increasingly more positive, it attracts more and more electrons to itself. Eventually, it would begin to rob the plate of electrons, so there is a limit to the amount of increase in plate current which we can get by making the grid positive.

Negative Grid Voltage. Now suppose we reverse the potential on the grid, making it negative (see Fig. 16).

The grid will now repel electrons. If the negative voltage is high enough, all electron movement between the oathode and plate will be stopped; or, if the grid is made only slightly negative, the plate current will be decreased.

You can see, then, that a voltage applied to the grid eloment will control or vary the plate current. If the grid is made negative, the current decreases, while a positive grid causes a current increase. Should we apply an a.c. signal voltage between the grid and cathode, the alternate positive and negative half-cycles will produce upand-down variations in the plate cur-



FIG. 14. With no grid bias, an average number of electrons flow to the plate, while the rest form the space charge between the cathode and the grid.

rent—up when the grid is positive, down when it is negative.

▶ However, we're not looking for plate current variations: what we want is an amplified signal voltage. And we can get it very simply just by connecting a "load" R_L in the plate circuit, as shown in Fig. 17. The varying plate current then flows through resistor R_L , and the resulting voltage drop across R_L of course varies in exactly the same way as does current—which, in turn, varies in exactly the same way as does signal voltage on the grid. Thus, each variation in the grid signal voltage



FIG. 15. Making the grid positive greatly increases the number of electrons moving to the plate.

causes a similar variation in the voltage drop across $R_{\rm L}$. In other words, the signal voltage fed to the grid is reproduced as a similar signal voltage across $R_{\rm L}$.

▶ Notice—only a reproduction of the grid voltage, not the grid voltage itself, appears across R_L . Always keep this point firmly in mind. The grid voltage merely changes the plate current—it is the changing plate current which causes the signal voltage to be reproduced across plate load R_L .

The source of the plate current is, of course, the B supply—either a B battery or an a.c. operated power pack. None of the plate current comes from the signal source. All that the grid



FIG. 16. A negative grid bias reduces the plate current.

signal does is control the flow of current from the B supply through the tube. You may find it easier to remember this action if you consider the grid to be a kind of "electrical valve" which controls the flow of plate current, but has nothing to do with supplying the plate current. (In fact, British technicians call tubes "valves" precisely because of this action.)

Now, is this signal voltage developed across R_L the voltage we want? Remember, we started out with the idea of somehow using a tube to amplify a small signal voltage—that is, we wanted to produce a voltage which would have the same general form as our signal voltage, but would be considerably greater in size.

We know that the voltage across $R_{\rm T}$ has the same form as the signal voltage fed to the grid-it goes up when the grid voltage goes up, down when the grid voltage goes down. The only question is whether the voltage across $R_{\rm L}$ is bigger than the signal voltage. As you know from Ohm's Law, the voltage drop across $R_{\rm L}$ is equal to the current flowing through it (the plate current) times its resistance, or $E = I \times R_L$. This equation shows us that if $R_{\rm L}$ is large, even a fairly small plate current through it will produce a large voltage drop across it. And, by the same token, a fairly small change in plate current will produce a large change in the voltage drop across $R_{\rm L}$.

Now, you learned a little earlier that the grid voltage is very effective in controlling the plate current: in other words, even a small change in grid voltage will produce a considerable change in plate current. And from what we said in the preceding paragraph, this change in plate current will produce a large voltage change across $R_{\rm L}$ (provided the resistance of $R_{\rm L}$ is large). Thus, a small signal voltage change fed to the grid will produce a large signal voltage change across R_{L} —so we have the *am*plified signal voltage we want.

▶ Where does this amplified signal voltage come from? Obviously, it is supplied by the only voltage source



FIG. 17. The positive and negative alternations of the a. c. signal cause the plate current to vary up and down, producing a similar variation in the voltage across R₁.

in the plate circuit—the *B* supply voltage. The grid voltage therefore controls the amount of the *B* supply voltage which is developed across load resistor R_L . This means that the maximum signal voltage variation we can get across R_L is equal to the total voltage furnished by the *B* supply. (Actually, we can never get this much of a variation, since a considerable amount of the *B* supply voltage is dropped across the tube itself.)

THE Eg-Ip CURVE

There are limits which must be placed on the operation of triode tubes. For one thing, the amount of signal which can be applied to the grid is limited; if too much is applied, the resulting changes in the plate current will not be exact "carbon copies" of the applied signal. We then say that the tube is "distorting" the signal. Of course, we do not have to worry about this when handling small signals, but, after several stages of amplification, the signal is built up to such proportions that tubes capable of handling large grid voltage variations must be used.

To understand these limits better,

let us study a curve showing the relationships between the grid voltage and the plate current. This curve is called the grid voltage-plate current curve or the "grid-plate characteristic" curve. Radiomen generally abbreviate these terms to $E_g - I_p$ curve or $E_g - I_p$ characteristic.

The typical $E_{\rm g}$ — $I_{\rm p}$ curve shown in Fig. 18 has a shape similar to that of the other curves we have studied. However, we have both positive and negative values of grid voltage, so we must insert the vertical line o-x, representing zero grid voltage (the grid connected to the cathode with no voltage between these elements). Then, the distance along the horizontal line to the left of line o-x represents increasingly negative grid voltage values, while the distance to the right represents positive values.

Let us start with a highly negative grid voltage, applied in the manner shown in Fig. 16. We represent this voltage by the distance from the o-xline to point A in Fig. 18. With this grid voltage, there is no plate current flowing.

Now, suppose we gradually reduce this negative voltage. When we get to the value represented by point Bplate current will begin to flow (inflicated by the fact that the curve begins here). Point B is known as the cut-off point, because it represents the negative grid voltage which cancels the effects of the plate voltage at the cathode. Any grid voltage more negative than this will cut off the plate current.

As we reduce the negative voltage further, the plate current increases. After getting past the bend or "knee" of the curve at point C, the plate current increases almost in step with the grid voltage change, causing the curve to be nearly a straight line between C and D. Then, when we pass the zero grid bias line o-x and make the grid increasingly more positive, we find the curve begins to flatten out toward E.

At the same time, as the grid is made increasingly more positive, grid current has begun to flow (shown by the dotted line). The positive grid is attracting electrons to itself. The grid current increases as the grid is made more positive; in effect, the grid robs the electron stream and thus reduces the number of electrons which we might expect to go to the plate.

Usually we do not desire grid current flow at all and do not desire to have the tube operate on any curved part of its characteristic when we are trying to amplify. (As we shall learn later, a curved tube characteristic will cause distortion.)

Grid Bias. To avoid grid current, the grid is kept negative at all times by a d.c. voltage connected between the grid and cathode in the manner



FIG. 18. This Eg—Ip curve is very important. It shows how the plate current varies with different grid voltage values.

shown in Fig. 19. (This voltage, commonly called a C voltage, is shown being furnished by a battery C in this figure, but other sources—which you will learn about later—are often used.) The value of this C voltage is chosen so that it causes the tube to operate somewhere near the middle of the straight portion of its E_g — I_p characteristic (point F in Fig. 20). This fixed grid voltage is called a bias voltage, a name which comes from the fact that it *influences* or *controls* the initial operating plate current. For example, a bias voltage which makes the tube operate at point F on the characteristic in Fig. 20 sets the initial plate current at the value M.

After the point at which the tube operates on its characteristic has been set by applying a bias voltage, we can then apply an a.c. signal voltage to the grid. This a.c. voltage alternately adds to, and subtracts from, the grid bias voltage—that is, instead of the grid swinging negative and positive, it becomes alternately more negative and less negative.

▶ For example, if the bias voltage is -10 volts, and we apply an a.c. signal having a peak value of 5 volts, then the grid voltage will change alter-



FIG. 19. Grid current flow is prevented by a negative bias.

nately from -5 to -15 volts. The grid voltage will be a pulsating d.c. voltage, since it will be a mixture of d.c. and a.c.

As Fig. 21 shows, these changes in grid voltage cause alternate increases and decreases in the plate current, just as any a.c. signal would. In this figure, point F represents the operating point of the tube as set by the bias voltage. When the grid voltage is varied by the a.c. signal represented by 1-2-3-4-5-6, the operating point must follow the instantaneous grid voltage up and down the operating curve between points F, C and D. Thus, when the grid bias is reduced by the signal swing from 1 to 2, the grid is made less negative, so that the operating point moves F to D, permitting an increase in plate current from 7 to 8. Similarly, during the swing from 2 to 3, the grid is made



FIG. 20. Point F is the operating point, which is fixed by the bias voltage.

more negative, so that the operating point moves from D to C, resulting in the plate current falling from 8 to 9. Similarly, as the signal voltage moves from 3 to 4 to 5 to 6, the plate current varies from 9 to 10 to 11 to 12. Hence, the a.c. grid voltage 1-2-3-4-5-6 produces the plate current variation 7-8-9-10-11-12.*

▶ Should the bias voltage change during operation, then the plate current would be forced to follow this change, producing variations which are not caused by the original signal. The bias supply must therefore furnish a steady d.c. voltage so that variations in the plate current will be caused only by the signal voltage.

^{*} A special data sheet on graphs will come to you with your graded answers for this lesson. This extra information was prepared as a part of the NRI Consultation Service, and there is no charge for it. This data sheet should be read carefully, as we feel its information will be particularly helpful with future lessons. However, remember that you don't have to draw graphs; you are expected only to be able to read them.

▶ Furthermore, the signal must not exceed the bias, because if it does, it can force the grid to become positive or can force the tube to operate over a curved section of its characteristic. Also, the bias must place the operating point properly on the straight portion of the characteristic curve.

As long as the tube operates along the straight-line portion of its characteristic curve, and as long as the a.c. voltage applied to the grid is not large enough to make the grid positive at any time, the plate current variation will be exactly like the grid voltage variation and will therefore produce voltage across the load resistor which is similar in form to the grid signal voltage. However, too high a signal voltage or operation over a curved portion of the characteristic will result in distortion. For example, in Fig. 22. the grid bias is too highly negative, so the operating point M is too near the lower bend of the curve. The curvature of the tube characteristic causes the bottoms of the plate current pulses to be chopped off, so the curve A-B-C-D-E-F-G is distorted (that is. it is not the same shape as the grid



FIG. 21. The manner in which the grid signal variations increase and decrease the plate current is shown here.



FIG. 22. If the operating point moves too far down on the curve, the plate current variations do not duplicate the entire signal.

signal). For perfect reproduction of the grid signal, the plate current pulses would have to follow the curve A-B-H-D-I-F-J.

THE TUBE AS A RESISTANCE

From what we've said about triodes, you can see that the kind of voltage we apply to a grid makes a vast difference in the performance and usefulness of a tube. If we apply only d.c. voltage (that is, bias voltage) to the grid, then the tube merely allows a d.c. plate current to pass through its plate circuit. A tube used under such conditions is nothing more than a resistor which limits the amount of d.c. flow from the B supply. When no a.c. signal is involved, we call the tube resistance its d.c. resistance. If we put a load resistor in the plate circuit, part of the voltage developed by the B supply will be dropped across the tube, the rest will be dropped across the load resistor. We can change the amount of current flow through the tube by changing the value of the bias voltage. The relative amounts of the battery voltage which are dropped across the tube and across the load resistor will change then.

If we increase the plate current, there will be a larger IR drop across the load resistor, and therefore a smaller drop across the tube. If we decrease the plate current, the IR drop



FIG. 23. Inserting an a. c. voltage in the plate circuit causes an a. c. variation in the plate current. The meter M measures only the a. c. portion of the current here.

across the load resistor will also decrease, and more of the *B* supply voltage will be dropped across the tube. In effect, a *triode* tube acts like a *variable* resistor, whose resistance we can change by changing the grid voltage. This is one way of looking at a tube—considering it as a variable resistor in series with a load resistor and a *B* supply. The resistance varies according to the signal variations, thus reproducing similar variations in the load voltage.

Unfortunately, such a picture of tube operation leads into difficulties when we try to calculate the *amount* of resistance variation, as this is a quite different value from the *d.c. plate resistance*. For this reason, we use another method of representing the a.c. operation of a tube, which can be used to compute the amount of a.c. voltage produced across a plate load resistor. This representation is called the "equivalent tube circuit." But before we go into it, we must learn the meaning of two useful new terms —a.c. plate resistance and amplification factor—which are always part of any discussion of the a.c. operation of a tube.

A.C. Plate Resistance. The a.c. plate resistance is the opposition which a tube offers to the passage of a.c. through it. It can be measured with a circuit like that shown in Fig. 23. In this circuit, we first apply normal voltages to the filament, grid, and plate of the tube, then add a known a.c. voltage to the d.c. voltage which is already applied to the plate. The addition of the a.c. to the d.c. plate voltage makes the total voltage applied to the plate alternately larger and smaller, which in turn makes the plate current alternately larger and smaller. The plate current, in other words, becomes a pulsating direct current-which, as you know, is a steady direct current to which an alternating current has been added. We can measure the a.c. part of the plate current with the a.c. meter M.

Next, we divide the amount of a.c. voltage applied to the plate by the amount of a.c. plate current produced. This gives us the *a.c. plate resistance* of the tube—the opposition which the tube offers to the flow of a.c. through it. (This is NOT the same value as the d.c. plate resistance of a tube, which is what we would get if we divided the d.c. voltage applied to the plate by the d.c. plate current. D.C. plate resistance is very seldom mentioned in radio work.)

Whenever you see the term "plate resistance" or the symbol r_p (sometimes R_P) used—in a tube chart, for example—you can be sure that a.c. plate resistance is meant.

This a.c. resistance will change if the filament voltage, the d.c. grid voltage, or the d.c. plate voltage is varied, so each of these voltages must be specified when we state the a.c. plate resistance of a tube. This is done in listings of tube characteristics in tube charts. The values of voltages given in these listings are generally those which would be used in some typical application of the tube concerned.

Amplification Factor. We have just seen that an a.c. voltage introduced in the plate circuit will cause an a.c. plate current, and we learned earlier in this lesson that an a.c. grid voltage will likewise cause an a.c. plate current. (In each case, of course, the a.c. plate current produced is mixed with the d.c. plate current which the B supply and the d.c. bias cause to flow, so the total plate current flow is a pulsating d.c.)

▶ One thing further we have learned: A grid voltage is more effective in controlling the flow of plate current than is an equal plate voltage. But we still have not learned any way of telling how much more effective the grid voltage is. That is what the amplification factor of a tube tells us.

We could find the amplification factor of a tube with a circuit like that



FIG. 24. The amplifying ability of the tube can be determined with this circuit.

in Fig. 24. First, we would turn off the a.c. source in the plate circuit, then apply a known a.c. voltage to the grid and measure (with a.c. meter M) the a.c plate current produced. Next, we would turn off the a.c source in the grid circuit, and would turn on the a.c. source in the plate circuit. We would adjust this a.c. plate voltage until we got the same a.c. plate current as the grid produced. Finally, we would divide the value of this a.c. plate voltage by the value of the a.c. grid voltage we used. The answer would be the *amplification factor* of the tube.

Thus, we can define the amplification factor of a tube as the ratio of the a.c. plate voltage to the a.c arid voltage necessary to produce the same a.c. plate current. This gives the relative effectiveness of the plate and grid voltages. For example, if we find that an a.c. voltage of 10 volts in the plate circuit is needed to produce as much a.c. plate current as an a.c. voltage of 1 volt in the grid circuit will produce. the amplification factor of the tube is 10 $(\frac{10}{1} = 10)$. In other words, the grid voltage is ten times as effective as the plate voltage in controlling the plate current. (The amplification factor is assigned the symbol "u," the Greek letter "mu," pronounced "mew.")

The amplification factor of a tube depends upon the physical structure of the tube—the relative spacing of the grid to cathode, compared to the plate-to-cathode spacing, and the actual grid structure. It is therefore relatively constant as long as the tube is not damaged in any way that would jar the elements from their proper positions.

Equivalent Tube Circuit. Now that we know what a.c. plate resistance $(R_{\rm P})$ and amplification factor (μ) mean, we are ready to take up the equivalent tube circuit which is so helpful when we want to learn what a.c. output to expect from a tube. This circuit is shown in Fig. 25. It is called an equivalent circuit because, as far as a.c. is concerned, it behaves just like a tube connected to a load. In other words, we transfer to the plate circuit an a.c. signal equivalent to an amplified grid voltage, then consider only the a.c. resistance of the tube.

Notice-in this circuit we replace the tube by an a.c. generator and a series resistor, and connect these two across the load $R_{\rm L}$. The generator has a voltage output equal to μ times e_{π} (e, is the a.c. grid signal voltage) and the series resistor has the value $R_{\rm P}$. The fact that these two elements are the equivalent of a tube (as far as a.c. is concerned) illustrates two important things: 1, that the total a.c. signal voltage produced in the plate circuit is equal to the grid signal voltage multiplied by the amplification factor of the tube: and 2, that part of this plate circuit signal voltage is always dropped within the tube across the a.c. plate resistance of the tube.

You can see at once why it is important to know the μ and $R_{\rm P}$ of an amplifier tube. The amplification factor μ tells us immediately how much the tube amplifies the grid signal. If a tube has a μ of 10, then the total signal voltage produced in the plate circuit will be 10 times the signal voltage fed to the grid.

The a.c. plate resistance $R_{\rm P}$ lets us determine how much of this amplified signal we can actually use. As you see from Fig. 25, R_P and the load resistance $R_{\rm L}$ form a voltage divider. Part of the amplified signal voltage is dropped within the tube, across $R_{\rm P}$; the rest of it is dropped outside the tube, across $R_{\rm L}$. Naturally, the only part of the signal voltage that is useful to us is the part dropped across $R_{\rm L}$, because this is the only voltage we can feed into another circuit. The signal voltage dropped inside the tube (across $R_{\rm P}$) is wasted; the power it develops is dissipated in heating the plate.

Naturally, then, the value of load resistor R_L is very important in deciding how much useful amplification

of the signal voltage we will get. If $R_{\rm L}$ is zero, we will get no useful amplification at all; if it is very much larger than $R_{\rm P}$, we will get a useful amplification almost as large as the μ of the tube. To distinguish between the useful signal amplification we get out of a tube circuit and the total signal amplification (μ), we call the former the stage gain. The stage gain



FIG. 25. The tube plate resistance and the load divide the a. c. voltage according to their values.

is always, of course, less than the μ of the tube: in fact, it is equal to μ times the ratio of the plate load to the sum of the plate load and the a.c. plate resistance. Or, in equation form,

Stage gain =
$$\mu \times \frac{R_{\rm L}}{R_{\rm L} + R_{\rm P}}$$

This equation again shows us that the larger $R_{\rm L}$ is with respect to $R_{\rm P}$, the greater the useful signal voltage amplification we can get out of the tube circuit. However, there are practical limits to the value of $R_{\rm L}$; a value five to ten times the plate resistance gives about all the voltage gain possible under normal conditions. We will study this problem more when we take up amplifying stages.

▶ While on this subject, we might mention that sometimes maximum voltage gain may not be as desirable as maximum power output. For example, it takes *power* to operate a loudspeaker; transmitters depend on power increases from stage to stage, etc. As we shall learn in a later lesson, power requirements affect the value of the load chosen and the tube design.

Remember—this equivalent tube circuit applies only to the a.c. signalamplifying operation of the tube. It has nothing to do with the d.c. voltages as applied to the tube (except that these voltages partially determine the R_{P} -and, in some special tubes, the μ -of the tube). You will meet the equivalent tube circuit many times more, both in your N.R.I. Course and in the literature of the Radio profession, because it is the most direct way to determine stage gain.

Multi-Element Tubes

We could make up any radio circuit with only the diode and triode tubes we have so far discussed. It would be impractical to do so, however, because many other types of tubes have been developed which are far more efficient for some radio uses than are diodes and triodes. We'll describe these other types briefly now, to give you an idea of their operations and uses; you'll learn all about them in later lessons of your Course.

SCREEN GRID TUBES

One important drawback of a triode, when used for radio frequency amplification, is the capacity between the grid and plate elements of the tube. These elements are conductors, separated by a dielectric (the vacuum), so that they make up a capacity (see Fig. 26). As you will learn in a later lesson, an undesirable feedback of energy occurs through this capacitive path, which causes r.f. amplifiers to become useless unless special circuits are used to cancel the effects of the feedback.

Tube manufacturers decided to improve the tube itself to eliminate this effect as much as possible. They brought out the screen grid or tetrode (four-element) tube, which has another grid located between the original grid and the plate, as shown in Fig. 27. To distinguish between these grids, the original triode grid is now called the control grid, while the second is called the screen grid. In structure, the two grids are somewhat similar.

The addition of the screen grid causes an enormous decrease in the capacity between the control grid and the plate, and so almost eliminates the undesirable feedback. (Later lessons will explain how this is accomplished.)

A positive voltage (with respect to the cathode) is always placed on the screen grid; this voltage is usually somewhere between half and full plate voltage. Since the screen is considerably closer to the cathode than the plate is, its voltage has far more effect on the electron stream than the plate voltage has. In other words, the screen arid voltage and the control grid voltage are the important factors which determine how many electrons pass through the tube. The plate voltage acts principally to collect the electrons pulled through the tube by the screen grid voltage, and does not have a great deal of effect on the number of electrons pulled through. fact, if we keep the control grid and screen grid voltages steady, we can vary the plate voltage considerably without changing the plate current much.

Amplification. You will recall that the amplification factor of a tube is the ratio of the effectiveness of the grid voltage to the effectiveness of the plate voltage in controlling the plate current. In a screen grid tube, the control grid voltage is just about as effective in controlling the plate current as it is in a triode, but, as we have just seen, the effectiveness of the plate voltage is considerably re-This means, therefore, that duced. the amplification factor of a screen grid tube is much larger than the amplification factor of a triode. Actually, the screen grid tube may have a μ between 100 and 600—while most triodes have a µ of 10 or less.

► However, as you learned in your study of the equivalent tube circuit, the amplified signal (μe_g) is divided between the tube a.c. plate resistance and the load resistance. The screen grid tube, unfortunately, has an ex-



FIG. 26. The grid-to-plate capacity is troublesome in many applications.

tremely high plate resistance. In fact, it is not usually possible to bring the load resistance anywhere near the plate resistance value, so the useful amplification (stage gain) of a tetrode is considerably less than its amplification factor. Even so, a stage gain of 100 is not impossible, which is certainly a much higher stage gain than any triode can give.

PENTODE TUBES

The screen grid tube has the important advantages over a triode of having negligible grid-to-plate capacity and far higher gain. However, it has disadvantages when handling large signals.

Because the screen grid (which accelerates or speeds up the electrons



FIG. 27. The screen grid reduces the gridto-plate capacity.

in their journey toward the plate) is fairly close to the cathode and has a high voltage, electrons travel much faster in a tetrode than they do in a triode. As a result, when they strike the plate, there is a much greater tendency for these speeding electrons to knock other electrons loose from the plate. This effect is known as secondary emission.

These secondary emission electrons fly off the plate. Many are attracted right back to the plate by its highly positive potential, but some go from the plate to the screen grid, which is also positive. This produces an undesirable current flow within the tube, and produces serious distortion if left unchecked. The effect is particularly noticeable as larger signals are being handled.

The *pentode* (five-element) tube shown in Fig. 28 solves this problem neatly. This tube has a third grid (a relatively coarse one, with only a few turns of wire) inserted between the screen grid and the plate. This new grid is usually connected directly to the cathode, and is at the cathode potential—that is, it is negative with respect to the screen grid and the plate. This new grid is called a suppressor grid (or sometimes a cathode grid, because it is connected to the cathode).

The operation of the tube is somewhat similar to that of the screen grid tube, in that the screen grid speeds up the electrons toward the plate. However, the suppressor grid, which is at cathode potential, slows down the electrons somewhat when they have passed the screen grid; it also reduces the velocity with which they hit the plate. (Because the suppressor is so coarse in structure, it does not interfere with the number of electrons passing through the screen to the plate.) Even so, some electrons strike the plate with sufficient velocity to knock



FIG. 28. The extra grid in the pentode tube reduces the effects of secondary emission.

out others, which then move into the space between the suppressor grid and plate. There they are subjected to two forces—the repelling force of the negative (cathode potential) suppressor grid and the attracting force of the positive plate. This double force makes practically all of them return at once to the plate.

BEAM TUBES

The beam power tube is another ingenious solution to this problem of secondary emission electrons. In effect, the electrons are made to act as their own suppressor grid. As Fig. 29 shows, a beam power tube has two small additional plates between the screen grid and plate. These small plates are connected to the cathode



Courtesy R.C.A. Mfg. Co., Inc.

FIG. 29. The electron "beam" provides its own suppression, forcing secondary emission electrons back to the plate.

just like a suppressor grid, and so repel electrons. This forces the electrons to "bunch up" in the space between these plates and flow in two concentrated streams or "beams" between the cathode and the true plate.

This concentrated electron flow prevents secondary emission electrons from reaching the screen grid, because any secondary electrons emitted from the plate immediately encounter the electron beams and are repelled back to the plate.

SPECIAL TUBES

The pentagrid converter (shown schematically in Fig. 30) is an example of a tube designed for a special purpose. The tube gets its name from the fact that it has five grids (penta means five), and is used as a frequency converter tube. We'll go into details of its use in later lessons. Obviously, however, with all these grids we can get many different controlling actions on the plate current.

▶ In the early days of radio, the amplification of a single tube was quite limited, so the only way of get-



FIG. 30. A pentagrid converter tube.

ting more amplification was to have more tubes. As a hang-over from those early days, many people still believe that the more tubes, the better the radio. This has resulted in radio manufacturers at times using separate tubes, when they could have lowered costs somewhat by using one or more of the dual tubes available.

 \blacktriangleright For example, there are many applications which require the use of two diode tubes in the same circuit.



FIG. 31. Dual diodes.

It is logical to place both these diodes in a single bulb. Naturally, since a dual tube of this kind needs only a single bulb and a single base, it is somewhat cheaper than two single tubes and permits the use of a smaller chassis.

The schematic symbols for typical dual diodes (double diodes) are shown in Fig. 31. Some applications require separate cathodes as shown in Fig. 31A, while others permit combining the cathodes as shown in Fig. 31B. ► Another popular combination is. that of the diode and triode tubes. The detector in most modern radios is a diode tube. Its output feeds into an audio tube, which is usually a tri-Another diode is desirable at ode. the same point in the circuit for control purposes, so very often all three are combined in modern radios into one dual diode-triode tube. Fig. 32



FIG. 32. Two diodes and a triode in a single envelope.

shows the schematic symbol for this tube. The same cathode is used for the triode and for the two diodes.

• Other similar combinations have been made. You will sometimes find together two triodes, a triode and pentode, two diodes and a pentode, and other similar combinations. Although these combination tubes are mounted in the same envelope, each section is generally independent of the others (except that they sometimes use the same cathode). You can always consider them to be separate tubes as far circuit operation is concerned. as We'll take up all these special tube types as we come to their uses in later lessons.

Special Purpose Tubes

There are other tubes designed for special purposes. Let's treat them briefly here, so you'll be familiar with their construction.

GAS FILLED TUBES

So far, we have been speaking entirely of vacuum tubes, which have as much air and gases removed as is economically possible or desirable. In a vacuum type tube, we do not want any gas ions to interfere with the electron movement. However, there is a special class of tubes, particularly diodes and triodes, in which a certain amount of gas is deliberately introduced.

These tubes are manufactured and the gases are evacuated in the same way as the ordinary vacuum tubes. Then, a definite, controlled amount of some particular gas is introduced in the tube to get some special action.

One result of this introduction of gas is a remarkable reduction in the plate resistance of the tube. When electrons move from the cathode to plate, they strike many of the relatively large gas molecules which are between the cathode and plate. The speed of the electrons is sufficient to knock other electrons out of these molecules. These extra electrons travel onward to the plate, along with the original electrons.

The heavy gas ions that remain are now positively charged (they have lost electrons). As a result, they move toward the cathode. When these positive ions get near the cathode, they combine with an equal number of the electrons in the electron cloud. If just the right amount of gas is used, practically all the electrons in the electron cloud will thus be removed from the cloud. In fact, the cloud will no longer exist, so all of the electrons leaving the cathode will go directly to the plate. This means a high plate current will be developed for a relatively low plate voltage.

Gas type diodes are found mostly in power supply systems, where their low plate resistance results in a smaller loss of power within the tube. The supply must be carefully designed to prevent the plate current from rising above a certain critical value, however, or too many heavy ions will be formed, with the result that the cathode surface will be bombarded by the heavy particles and the oxide coating knocked off.

► Gas type triodes find special uses in industrial control equipment. These tubes are designed so that the grid merely acts as a trigger, starting the plate current flow at a certain predetermined value of grid voltage and then losing all control of it. This particular action is very desirable for certain control purposes, making it possible for a low-power circuit to turn on a high-power circuit.

PHOTO-ELECTRIC CELLS

In the beginning of this lesson, we mentioned that the thermionic (heating) method was only one of the ways in which electrons could be released from a cathode. Another method is used in one type of photo-electric cell.

Certain rare earth metals have the property of releasing electrons from their surface when light falls on the surface. (This is called the "photoelectric effect.") If we use one of these metals as a tube cathode and expose it to light, electrons will escape from it. Then, if a positive element is nearby, we will get an electron flow from this photo-electric cathode to the positive element, just as in our thermionic diode tube. The amount of current depends on the number of electrons released, which in turn depends on the amount of light.

A typical photo-cell is shown in Fig. 33. The large "plate" is actually the cathode, as it is the surface on which light falls. As electrons are emitted from this surface, they travel to the thin wire, which is positively charged with respect to the cathode.

The photo-electric principle finds many uses in industrial control equip-



FIG. 33. A typical photoelectric cell.

ment. For example: Moving objects can interrupt a light beam shining on a photo-electric cell, thus momentarily cutting off the current through the cell and causing counting devices to operate.

▶ The television camera tube, used to pick up the image at the transmitter, is also a photo-electric device. There are many types of these tubes. One type is really a combination of thousands of miniature photo-electric cells mounted together on a plate. When exposed to a scene, the various photoelectric cells emit electrons according to the light values of the tiny segment of the scene to which each is exposed. It is possible to pick up these smallcharges with a scanning device and transmit them as elements of a television scene.

COLD-CATHODE TUBES

There have been several types of dual diodes (known as "cold-cathode" tubes because their cathodes are not heated) developed for power packs.

These tubes are gas-filled. When the plate element is sufficiently positive with respect to the cathode, the gas will ionize, as electrons will be pulled out of the gas molecules toward the plate. Then, the heavy positive ions will bombard the cathode surface and knock electrons loose to replace those taken from the gas by the plate. The gas molecules thus re-form (become un-ionized) and the action repeats itself continuously as long as the proper potentials are maintained between the plate and cathode.

This action is reversible, since reversing the applied voltage would cause the plate and cathode to be interchanged. If it is desired to use such a tube as a rectifier (in which current flows only one way), one surface can be treated to release electrons more readily than the other, or the element intended to be the cathode can have a far larger surface area than the plate. This makes the tube conduct better one way than the other. Rectification is not perfect, as some current can still flow in the reverse direction, but it is satisfactory for some applications.

Tube Classification Systems

There are hundreds of different vacuum tubes, so it is absolutely necessary to have some means of telling them apart. Of course, one obvious difference between tubes is the number of elements in the tube. Hence, we have diodes, triodes, tetrodes, pentodes, and others. However, these classifications are too broad to be very useful in telling tubes apart, for there are many tubes of each of these types which differ in voltage requirements and other characteristics.

Thus, we have triode tubes with low amplification factors and triode tubes with high amplification factors. There are kinds of low-amplification tubes which differ greatly in plate resistance and in the voltages required on the filaments and other elements. Similarly, high amplification triodes are made in models requiring many different filament, grid, and plate voltages. Some of these tubes are intended to operate only from battery supplies, while others will operate with a.c. on the filament.

Furthermore, tubes vary greatly in size. There are tiny tubes for hearing aids and super-portable receivers, medium size tubes for ordinary small receivers, and full size tubes for standard radio receivers. Naturally, the basic operation of the tube is not affected by its size. However, it frequently happens that a regular size tube will not fit into a radio designed to use small tubes.

Filament Voltages. Thermionic tubes may be classified according to the filament voltage which they are designed to use. Plate or grid voltages may be abnormal for a particular tube without immediate or great damage, but if the filament voltage is 'too high, the filament will burn out. It is important that the filament voltage requirements be met. However, many tubes use the same filament voltages, so classification by filament voltage is (like classification by number of elements) too broad to be very useful.

Use Classification. Tubes may be classified according to their use, or rather, according to the use for which they were originally designed. Thus we have rectifiers, detectors, mixers, oscillators, r.f. and a.f. amplifiers, and power amplifiers. Often, however, it is possible to use one tube for several different purposes, so a given tube may have two or more usage classifications. Some time ago it was decided that the simplest identification system was to assign each tube type a number and to list tubes by their numbers in charts or tables of tube characteristics. We shall go into this system shortly—but first, let's learn something about how tubes differ in outside appearance.

TUBE BASES AND SOCKETS

Unlike other devices used in radio and electronic apparatus, the radio tube is a délicate part giving limited hours of service. Most manufacturers of radio tubes guarantee their products for one thousand hours. While tubes in general give much longer service than this, it is not impossible for a tube filament to burn out during the first few hours of its use. For this very practical reason, tubes must be built in such a way that they can easily be removed for testing and replacing. Hence, radio receiving tubes have bases with prongs or pins which serve as the terminals of the tube elements. These bases plug into sockets having spring contacts which bear against the prongs when the tube is in the socket. The circuit wiring is connected to the socket contacts, and thus to the tube elements when a tube is plugged in.

It is, of course, necessary to make sure that a tube will fit into its socket in only one way, so that the socket contacts will always bear against the proper prongs. In older tubes, prongs of different sizes or different prong spacing arrangements were used to make the tubes fit just one way in sockets. Of course, this system means



FIG. 34. The pin arrangement of an octalbased tube, which also corresponds to the bottom socket connections.

a different socket is necessary for each tube base type.

▶ Most modern tubes have "octal" bases which will fit a universal octal socket. All prongs on this base are the same size and are equally spaced. An aligning key in the center of the tube base fits a slot in the socket in only one way, thus assuring insertion of the tube in only one way.

Fig. 34 shows the prong arrangement of an octal base. This same base is used on diodes, triodes, pentodes, etc.; if any prong is unnecessary for a particular tube, it is simply left off. This does not affect the positions of the other prongs; if prong 6 in Fig. 34 is omitted, all other prongs will still remain in the same positions and will still be designated by the same numbers.

Fig. 35 shows typical tube bases of both the old and the octal types.

► Certain sets need tubes which will not easily loosen in their sockets. (Auto sets and portables are examples.) The "loctal" base is the answer to this. It is similar to the octal base except that the centering key has a ridge, shown in Fig. 36, which snaps into a ring in the socket made of spring material and locks the tube in the socket. It is necessary to push a loctal-based tube sideways to release the key from the spring before it can be pulled out of the socket.

The prongs are exceptionally small on these tubes—really just projections of the connecting wires which come through the glass seal at the bottom of the tube. This eliminates the necessity for a tube base, as the glass seal positions the pins and the centering key is molded into the seal. These tubes offer advantages at ultrahigh frequencies in that the elimination of the base reduces the amount of leakage at these frequencies.



Courtesy Emerson Radio & Phonograph Corp.

The relative size of miniature tubes is shown here—four are held in the palm of a hand.

(Many materials are good insulators at low frequencies but become poor insulators at the frequencies used for television and f.m. broadcasting.)

▶ Hearing aids and the extremely small portables use miniature tubes, only ¾ inch in diameter and less than two inches high. Actually, these tubes have no true base; the glassbulb bottom is flattened and has a ring of holes. The prongs (really just connecting wires, like those of a loctal tube) come through these holes, which are placed at the proper points to fit the socket.

Top Caps. The first tubes had all connections made at the base. However, when the need for reduced grid-to-plate capacity was recognized, tubes were developed with a terminal in the form of a metal cap at the top of the glass or metal envelope, in addition to the prongs or pins. Connectpacity, so many tubes are now made with the control grid lead going to one of the prongs in the base. This is now possible because a special type of base construction is used to shield the electrode leads and prongs from each other and so reduce lead capacities. These tubes (called single-ended tubes because all connections are at the same end) permit all connections



FIG. 35. Representative tube bases, showing standard prong arrangements A—a 4-prong base; B—a 5-prong base; C—a 6-prong base; D—a large 7-prong base; E—a small 7-prong base; F—an octal base having only 5-prongs; G—an octal base having all 8 prongs.

ing the grid to the cap makes possible a maximum spacing (and therefore a minimum capacity) between the grid and plate leads. Connections are made to this cap (usually called the top cap or grid cap) by a topcap clip. A flexible wire lead connects this clip directly to some part in the chassis of the radio apparatus; the clip must naturally be removed before the tube can be removed from its socket.

Single-Ended Tubes. The top cap connection is not too desirable from a mechanical standpoint, as loose connections frequently develop. Tube manufacturers discovered a means of eliminating the top cap without a great increase in the grid-plate cato be made underneath the chassis, simplifying the wiring and eliminating long leads through the chassis to the top cap. These tubes have either octal or loctal bases, so that they fit the corresponding socket.

PRONG IDENTIFICATION

After base types and sizes of tubes were standardized, it became possible to number the prongs and socket clips







FIG. 37. These schematic symbols not only show which elements go to which prongs they also show the relative positions of the prongs. Notice that the octal bases may not have a full set of 8 prongs but when prongs are missing, the others maintain their original spacing and numbers. These connections are correct when viewing the BOTTOM of the base or tube socket but are proper just for CERTAIN TUBES. Always consult a tube chart to determine the connections on tubes.

so as to identify the element connections. A standard system was developed by the Radio Manufacturers Association (R. M. A.) for this purpose. Looking at the bottom of the socket, the numbers progress *clockwise* around the socket. Examples showing the element connections for several tube types are shown in Fig. 37. These numbers are in order when you look at the *bottom* of the socket, because this is the normal position for radio servicing.

Locating Filament Prongs. In 4. 6. and 7-prong tube standard bases. those two prongs having the highest and lowest numbers are made larger than the others, to insure that the tube is always inserted in its socket in the correct position. For example, pins 1 and 4 of a 4-prong tube would be larger than the others: these thicker prongs are always connected to the filament. In the standard 5-prong tube, all pins are of the same diameter but are so arranged that there is only one way of inserting the tube; pins 1 and 5, the filament prongs, are always close together and directly opposite a single pin. The positions of the filament prongs in octal tubes vary with different tubes, so it is generally necessary to refer to a tube chart for accurate information.

Locating Prong No. 1. You can locate any prong by remembering that the R. M. A. numbering system always progresses clockwise when you look at the bottom of the tube socket. In octal tubes, prong 1 is always in the clockwise direction from the aligning slot as you look at the bottom of the octal tube socket. Notice the examples given in Fig. 37.

TUBE TYPE NUMBERS

Since it is clearly impractical for busy radiomen to describe a certain tube completely when speaking of it. each tube has been assigned a number. In the early days of radio, when only a few different types of radio tubes existed, there was little confusion even though numbers were assigned to new tubes in haphazard fashion; tubes such as the 01A, 12, 26, 71A, 56, 58, and 24 are examples. Notice that these numbers give no indication of the characteristics of the tubes. As the number of different types of radio tubes increased, it became increasingly more difficult to recognize



Courtesy Hyprode Sylvenia Corp

INSULATING DISC GLASS BEAD SEAL FERNICO EYELET

-SOLDER TOP CAP

A (above). The assembly of a typical all-metal tube, the type 6A8 pentagrid converter, can be traced in the above photograph. Metal exhaust tubing and The above photograph. Mictal exhaust tubing and Fernico metal eyelets are welded to metal header shown at upper left, then glass beads with wire leads are threaded through eyelets and fused to them. Electrodes are welded to projecting leads, metal envelope is welded to header, tube is evacu-ated and sealed off, octal bakelite base and top cap are argumed into noticing and tube is prayed accouare crimped into position and tube is sprayed, com-pleting job. A completed tube with half of envelope cut away is also shown.



tees Battensi Unten Radio Oory

C

Courtesy R G.A. Monafasturing Co., inc

35 World Radio History tubes. The following code, which is now standard for all new tubes, was developed by the R. M. A. to eliminate some of this confusion, as this code gives at least *some* clue to the tube type.

R.M.A. TUBE NUMBERING CODE

Each tube designation shall consist of a number (or digit), followed by a letter, which is in turn followed by another digit.

(a) The first numeral (or group of numerals) shall indicate the *filament* voltage in steps of 1 volt, using 1 to mean any voltage below 2.1; 2 to mean 2.1 to 2.9 volts; 3 to mean 3.0 to 3.9 volts; 4 to mean 4.0 to 4.9 volts; 117 to mean 117.0 to 117.9 volts, etc.

(b) The last numeral shall designate the number of useful elements (filament, cathodes, grids, plates, etc.) which are connected by wire leads to prongs or the tube cap. The filament is here counted as one element, and the envelope of a metal tube (or a shield within a glass tube) is considered as one element.

(c) The letter between the numerals shall be a serial designation which will serve to distinguish between tubes having the same number of useful elements and the same filament voltage. Rectifiers will start with Z and work backward through the alphabet, while all other tubes start with A and work up through V of the alphabet. When all 26 letters of the alphabet have been used for a given combina-

tion of first and last numbers, the next 26 new tubes with these numerals will have the letter A ahead of the serial designation letter; succeeding groups of 26 tubes will have B, C, D, etc., ahead of the serial designation letter. Examples: 6AB5; 6AF6.

(d) The letter S ahead of the serial designation letter (following the first numerals) indicates a single-ended tube. Example: 6SK7. The letter G following the last numeral indicates an octal-base tube having a standard glass envelope instead of a metal envelope. Example: 6Q7G. The letters GT following the last numeral indicate an octal base tube having an extra-small glass envelope instead of a metal envelope. Example: 6Q7G.

▶ The foregoing system is primarily intended to be a logical system of assigning tube numbers in such a way that some useful information can be conveyed to the user. However, a tube chart must be used to find the filament current, plate, and grid voltages, plate current, amplification factor, prong connections, Tubes are listed in such charts etc. by their numbers, so you need merely look up the number of the tube in which you are interested to find its characteristics and recommended uses. Furthermore, tube charts give the tube base connections, so it is possible to find from them which prong is connected to which element. We will give more information about tube charts elsewhere in your Course.

THE N. R. I. COURSE PREPARES YOU TO BECOME A RADIOTRICIAN & TELETRICIAN (REGISTERED U.S. PATENT OFFICE) (REGISTERED U.S. PATENT OFFICE)

Lesson Questions

Be sure to number your Answer Sheet 8FR-4.

Place your Student Number on every Answer Sheet.

Send in your set of answers for this lesson immediately after you finish them, as instructed in the Study Schedule. This will give you the greatest possible benefit from our speedy personal grading service.

- 1. What two electrodes must every electronic tube have?
- 2. Will electrons flow through an electronic tube when the anode is *negatively* charged with respect to the hot cathode?
- 3. Does the A battery furnish the electrons that are emitted by a heated filament?
- 4. Suppose an a.c. voltage source is connected between the plate and the cathode of a diode thermionic tube. Will the resulting plate current be: 1, a.c.; 2, pure d.c.; or 3, pulsating d.c.?
- 5. Is a tube which has an indirectly heated cathode called a *heater type* tube or a *filament type* tube?
- 6. What is secondary emission?
- 7. What is the purpose of the control grid which is placed between the cathode and the anode of an electronic tube?
- 8. Does the plate current: 1. remain constant; 2, decrease; or 3, increase; as the C bias voltage is made more negative?
- 9. How many elements are there in a pentode tube?
- 10. What does the first numeral (or group of numerals) in the R.M.A. Tube Numbering Code indicate?

YOU HAVE AN AIM IN LIFE

When you enrolled as a student member of the National Radio Institute, you took the first step on your road to success and happiness. You now have a goal for yourself—you have an aim in life—you are looking forward to the sort of work you like, the sort of income you want, and the respect and admiration of your friends.

Keep your goal in mind. Never forget it for a moment. Of course you will have your moments of discouragement—we all have. But if yon make a thorough search for the cause of your discouragement, you will most likely find that you ate something which did not agree with you, or were kept awake last night by the neighbor's dog. Realizing this, you will put your lessons aside for the time being, and tackle them the next day with renewed vigor and a renewed determination to succeed.

Whenever you are tempted to neglect your studies, say to yourself: "I have a goal to reach and I'm going to reach it." Think how unhappy you would be if you did not have this goal. There is nothing as pathetic as a rudderless ship or a man without an aim in life.

Here's to success and happiness—your goal. J. E. SMITH.





Study Schedule No. 9

For each study step, read the assigned pages first at your usual speed. Reread slowly one or more times. Finish with one quick reading to fix the important facts firmly in your mind, then answer the Lesson Questions for that step. Study each other step in this same way.

1-2 This short section introduces the L-C circuit, and tells how it is different from the filter circuits you have studied so far.

7 2. Series Resonant CircuitsPages 3-9

By comparison with familiar circuits, you are shown how a resonant circuit of the series type functions. Here is a remarkable circuit, capable of stepping up voltage, due to the cancellation effects produced hy phase. Study and restudy this section. Be sure you understand how the series resonant circuit works hefore going on, as you will find this information valuable when you study r.f. circuits. Answer Lesson Questions 1, 2 and 3.

3. Parallel Resonant CircuitsPages 10-14 The simple act of putting the coil and condenser in parallel instead of in

series gives a completely different circuit action. Yet, hoth circuits will accept or reject signals if properly used. Answer Lesson Questions 4 and 5.

The resistance within a resonant circuit has much to do with its ability to select, as you learn from the section on the Q factor. There are also sections giving practical ways of distinguishing series and parallel circuits, and on tuning these circuits. Answer Lesson Questions 6 and 7.

☐ 5. How Resonant Circuits are UsedPages 19-23 More practical data-showing typical methods of putting resonance circuits to use selecting or rejecting signals. Answer Lesson Question 8.

There are four ways of connecting a voltage source to a resonant circuit. These methods of coupling and an introduction to inductive coupling are given here, again from the practical "how it works" angle. Answer Lessou Questions 9

- 7. Mail Your Answers for this Lesson to N.R.I. for grading.
- 8. Start Studying the Next Lesson.

and 10.

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HOW TUNED CIRCUITS FUNCTION

COUPLING RADIO CIRCUITS

Introducing Resonant Circuits

SEPARATING different frequencies which have become mixed is an important task in radio. An antenna picks up signals from many different stations and we want to pick out just one of these. Within the radio, we have cases where we prevent hum, correct the tone quality and do other desirable things by making the proper separation. As you learned in a previous lesson, coils and condensers can be used for this purpose. If we want to pass on to the load only frequencies below a certain value, we can use a coil between the source and load, or we can use a condenser in parallel with the load. To make the cut-off more sharp, we may use both.

Similarly, if we wish to pass on to the load only frequencies *above* a certain value, we merely reverse the connections. This time, we use a con-



Courtesy General Electric Co.

Resonant circuits are used to select the desired station from among all others which are broadcasting. When you tune a radio, you are varying either the L or the C of these circuits to make their reactances balance at the desired frequency.
denser between the source and load; we use a coil in parallel with the load; or use both.

These circuits are known as filter circuits — called *high-pass* filters if they allow only frequencies above a certain cut-off frequency to pass, and called *low-pass* filters if they allow only frequencies below the cut-off frequency to pass.

▶ Notice that these circuits use only a single coil or a single condenser, or else they use one in series and the other in parallel with the load. There is another combination of a coil and condenser in which both are used together. This combination acts in a different manner altogether, forming the basis for new groups of filters.

► As you will soon see, there are many instances where we do not want all the frequencies above a certain value, or all the frequencies below a certain value. Instead, we may want to pass a certain narrow band of frequencies and exclude all others above or below this particular frequency range. Or, we may want to reject a narrow band of frequencies, yet pass all others above and below this narrow band. We can get these actions by: 1, using a coil-and-condenser combination between the load and the source; 2, using the proper combination in parallel with the load; or 3, using the proper combination as the load. This combination is known as an "L-C" circuit (L is the symbol for inductance; C is the symbol for capacitance).

► This particular action occurs because of the manner in which the reactances of these parts vary with the frequency. As you know, an increase in frequency causes a higher inductive reactance, while the same increase in frequency causes a lower

capacitive reactance. In other words, if we increase the frequency applied to a coil-condenser combination, the reactance of the coil will increase while the reactance of the condenser will decrease. Hence, with any coil-and-condenser combination, there is some <u>fre-</u> quency at which the two reactances are exactly equal, as will be shown in this lesson. A very special action occurs at this frequency.

By choosing different coil and condenser values, we can "move" this action to a different frequency. This makes it possible for us to "tune" a radio receiver (by using variable coils or condensers) so that we can select the desired signals and exclude signals from other stations operating on other frequencies. The L-C circuit used for frequency selection is called a tuned circuit or a resonant circuit.

► Another important use for this characteristic of a resonant circuit is in the generation of radio waves. The fact that this particular combination of parts can be made to accept some one frequency and exclude others makes it possible to use this combination with vacuum tubes in such a way that radio frequency waves of a desired frequency are generated. Thus, the L-C circuit can be used in transmitters and in oscillators to produce any desired frequency, while other L-C circuits can be used to accept or reject a particular frequency from among many others.

► Before we can see how L-C circuits are used for these purposes, we must learn how they may be connected together and how they act at resonance. Therefore, in this lesson we will show first how the two main types of resonant circuits function, then discuss some of the ways in which they are used in radio.

Series Resonant Circuits

There are two distinct types of resonant circuits, each having its own peculiar behavior. In a series resonant circuit, the source of voltage, the coil, and the condenser are all in series. In a parallel resonant circuit, the source of voltage, the coil, and the condenser are all in parallel. We will consider series resonant circuits first. ▶ The easiest way for you to understand the important actions that take place in a series resonant circuit is to analyze several circuits in turn. We'll start with a circuit composed of an a.c. generator and a variable resistance; then we'll add an inductance: next, substitute a capacitance for the inductance; and finally, combine the inductance, capacitance, and resistance into a resonant circuit. At each step, we'll see what current flows in the circuit and what voltages exist. This step-by-step procedure will show you exactly what effect each circuit element has on a resonant circuit.

For our explanation, we'll assume that the a.c. generator in the circuit produces 110 volts with a frequency of 500 cycles. This frequency is chosen purely for convenience, because it allows us to form a resonant circuit with inductance and capacity values which are easy to handle in calculations. Actually, any frequency—even r.f.—could be used and would show us the same fundamental facts, if the parts values were properly chosen.

The Resistance Circuit. Fig. 1A shows a circuit with which you're already familiar. It is made up of a variable resistor—value set at 120 ohms—in series with the a.c. source. An animeter is put in the circuit, and a voltmeter is placed across the resistor to show the current and voltage that exist in the circuit. As you already know, all the source voltage will be developed across the resistance, so the voltmeter reads 110 volts. And Ohm's Law for this circuit tells us the current flowing will equal the voltage divided by the resistance, so our ammeter reads .92 ampere (actually .917 as it is $110 \div 120$, but the meter reading will be so close to .92 that we would accept this as our reading. Other similar practical values are used in this lesson.) You learned from your previous lessons that this current is in phase with the voltage.

The Resistance-Inductance Circuit. Now, suppose we add a 100millihenry (mh.) coil in series with the resistance; this will give us the circuit shown in Fig. 1B. The ammeter now shows a current flow of about .33 ampere. The reason for the decrease in current is that we have added the reactance of the coil to the circuit.

The voltage developed across the coil is shown by a meter to be 103 volts, while that across the resistance



FIG. 1. When either a coil or a condenser is used in series with a resistor, the circuit current is reduced, as we would expect.

is 39 volts—a total of 142 volts, which is considerably more than the 110 volts generated by the source. As you learned in your lesson on coils, *phase* causes this effect. The voltage across the coil is out of phase with the circuit current (the voltage *leads* the current by 90°), while the voltage across the resistor is in phase with the current.

The Capacitance-Resistance Circuit. Now, let's replace the 100-mh. coil with a 1-mfd. condenser as in Fig. 1C. At 500 cycles, the reactance of a condenser of this size is approximately the same ohmic value as that of the 100-mh. coil. (If you wish, you can check this fact by figuring the reactances of the coil and of the condenser by the methods given in earlier lessons.) We see at once that this is true when we measure the current and voltages in the circuit. The voltage across the condenser is 103 volts, and that across the resistor is 39—a total of 142 volts, exactly what we found in the resistance-inductance circuit. And the current is again .33 ampere.

As you know from your study of condensers, the voltage across the condenser is *out of phase* with the circuit current (the voltage *lags* the current by 90°) while the resistor voltage is in phase with the current.

THE RESONANT CIRCUIT

So far, the actions of the components of the three circuits we've analyzed have been familiar. If we combine the three parts, then, we will have a circuit with a condenser, a coil, and a resistor all in series with an a.c. source, and you might expect the current to decrease further.

But—something unusual happens! The voltages and the circuit current increase tremendously when compared with the values of Figs. 1B and 1C. As Fig. 2 shows, a current of .91 ampere flows, and the voltage across the resistor rises to 109 volts. These values are almost what they were in the simple resistance circuit shown in Fig. 1A. Evidently, as far as the resistor voltage and the circuit current are concerned, the coil and the condenser might as well not be in the circuit.

▶ See what happens to the condenser and the coil voltages! Both increase to 285 volts—well over twice the source voltage. The circuit current has increased to almost three times what it was in the circuits shown in Figs. 1B and 1C. As we would expect, the voltage drop caused by this current flow through the coil and condenser reactances has also increased almost three times—from 103 volts to 285 volts. Just what makes this increase possible?

There is a very simple reason for these increases. The condenser voltage is 90° behind the circuit current, and the coil voltage is 90° ahead of the circuit current. Therefore, the two voltages are 180° out of phase with each other, which means their effects are exactly opposite. And. since the voltages are equal in size and opposite in effect, they cancel one another completely so far as the *circuit* is concerned. A voltmeter placed between points X and Y in Fig. 2 would show no voltage. (That is.



FIG. 2. When both the coil and the condenser are connected in series and are resonant to the applied frequency, their reactances effectively disappear. The circuit current is now determined by the source voltage and the resistance. The voltage drops across both the coil and the condenser are large, and their values are the product of the circuit current multiplied by the respective reactances.

it would if the coil were a perfect inductance, without resistance. But any practical coil has a certain amount of resistance, which will cause a slight extra voltage drop across the coil that the condenser voltage will not cancel. This is the reason why the full source voltage does not appear across the resistor in Fig. 2—one volt has been dropped in the coil resistance.)

This canceling effect of the voltages across the coil and the condenser is always present to some extent in a series L-C circuit, whether or not resonance exists. That is, even when the coil and the condenser reactances are not equal, some cancellation will occur. If the condenser voltage is smaller than the coil voltage, it will cancel a part of the coil voltage, and so cause a slight increase in circuit current over that which would flow if the coil were used alone. The same condition holds if the coil voltage is smaller than the condenser voltage. But the canceling effect becomes extremely noticeable only when the coil and the condenser reactances are equal for the particular frequency fed into the circuit-that is, when the circuit is at resonance. Then we get the enormous increase in circuit current we found in the circuit of Fig. 2, where practically all that limits the current is the resistor.

Reducing R. Fig. 3 shows what happens if we reduce the size of the resistor in our resonant circuit. Here, the electrical sizes of the circuit components are the same as in Fig. 2, except that the ohmic value of the resistor is reduced from 120 to 50 ohms. The circuit current increases to 2.1 amperes—and the condenser and the coil voltages (the result of this current flow through their impedances) jump to 670 volts! This shows that the value of the resistance in a resonant circuit is very important in



FIG. 3. The voltages measured here prove that the circuit current in a resonant circuit depends only on the circuit resistance.

determining the amount of current flowing. Notice, however, that less voltage appears across the resistor in the circuit of Fig. 3 than was across it in Fig. 2—107 volts for the former as against 109 volts for the latter. This difference is caused by the increased voltage drop across the resistance of the coil when the current is increased.

We cannot, therefore, produce an almost infinite current or an almost infinite voltage across the coil or across the condenser by reducing resistor R to zero. The resistance of the coil will limit the current flow even if there is no other resistance in the circuit. However, if we used a coil with very low resistance and carefully removed as much resistance as possible from the remainder of the circuit, we could get a very high current in our resonant circuit (and so a very high voltage across the coil and across the condenser). Of course, in practice, the source voltage usually will range from a small fraction of a volt to just a few volts, so even at best the condenser or the coil voltage is small. However, in such a circuit, it is sometimes necessary to limit the source voltage so the current will not rise to such values that the condenser voltage ratings are exceeded.

Resonant Voltage Step-Up. One of the most useful effects in a series resonant circuit is the fact that a high voltage—much higher than the source voltage—can exist across the coil and across the condenser at the resonant frequency of the combination. This is called resonant voltage step-up. By connecting across either the coil or the condenser (but not across both), we can use the voltage across that part to operate another circuit. The resonant circuit in effect has "amplified" the signal to which it is tuned, but will not so treat other frequencies. Hence, the circuit "selects" particular frequencies by increasing these more than others. You will meet and use this effect constantly in your later work in radio.

Conditions in a Series Resonant Circuit. Let's repeat the substance of what you've just learned, so it will be firmly fixed in your mind. In a series resonant circuit at resonance we have:

1. Maximum circuit current, determined by the resistance in the circuit and the source voltage.

2. Maximum and equal voltages across the coil and across the condenser, determined in size by the circuit current and by their reactances.

3. The L-C part of the circuit (the section between points X and Y in Fig. 2) acts like a resistor of low ohmic value; this resistance is almost entirely concentrated in the coil.

► Remember—this condition of series resonance exists for the circuits we've been considering only at one particular frequency. We deliberately chose the values of inductance and capacity used so that their reactances would be equal at 500 cycles, the frequency of the source. Their reactances would not be equal at any other frequency, so, with the particular condenser and coil we have used, our circuit cannot be resonant at any other frequency. In other words, our circuit is tuned to resonance at 500 cycles.

VARYING L AND C

Let's see what would happen if we were to use other values of capaci-

tance or inductance in our circuit.

Varying C. Suppose we were to vary the value of C (by inserting condensers of different sizes) while leaving the source frequency, the coil, and the resistor just as they are in Fig. 2. We are particularly interested in the circuit current I, since it determines what the voltage across L or C will be.

Fig. 4 shows a plot of the circuit current I for various values of C. With condensers smaller than 1 mfd., circuit current I drops rapidly; the



FIG. 4. This graph shows how the current varies in a series resonant circuit when the value of capacity C is varied from 0 to 3 mfd. Note that at 500 cycles and with a 100 mh. coil, resonance and maximum current occur when the capacity of C is 1 mfd., making its capacitive reactance equal to the inductive reactance of L.

smaller the value of C, the lower the current. The reason is quite simple; reducing the electrical value of C increases its reactance above that of L, and that portion of this capacitive reactance which is not cancelled by L acts to limit current flow.

With condensers larger than 1 mfd., the ammeter reading likewise drops; increasing the electrical value of the condenser makes the reactance of Clower than that of L, with the result that a portion of the inductive reactance is left over to limit current flow. ▶ Incidentally, when we tune our series resonant circuit away from resonance, the circuit will act the same as that reactance which is the larger or is predominant. Lowering the value of C here makes the capacitive reactance the larger, and the series resonant circuit therefore acts like a condenser, while increasing C makes its reactance less and the circuit acts like an inductance.

Varying L. If the inductive reactance (instead of capacitive reactance) were varied by inserting various values of L in the series resonant circuit, similar results would be obtained, as it is the amount of reactance left over (not balanced out) which governs the current. This is



FIG. 5. Varying L instead of C produces this kind of graph. Resonance again is obtained when the coil and condenser reactances are equal.

shown by Fig. 5—a graph of the circuit current for various values of L.

Varying Both L and C. With a 500-cycle source, we know that resonance can be obtained with a 1-mfd. condenser and 100-mh. coil. Could resonance also be obtained at this same frequency with other values of L and C? Suppose we try using a 50-mh. coil and vary C. If we do, we will find that we get resonance (maxi-

mum current) with a 2-mfd. condenser. Now notice—multiplying together the values of L and C in our original circuit (Fig. 2) gives a figure of 100 (100×1) . We get exactly the same figure for a circuit with a 50-mh. coil and a 2-mfd. condenser $(L \times C = 50)$ $\times 2 = 100$). As a matter of fact, we would find, if we tried other values of L and varied C to bring the circuit to resonance at 500 cycles, that multiplying together the values of L in millihenrys and C in microfarads at resonance always will give the number 100 for a 500-cycle source (this figure is sufficiently accurate for practical purposes). The larger the value of L_{i} the smaller will be the value of Crequired for resonance.

▶ If we changed the source frequency to, say, 1000 cycles, then changed either L or C until resonance was reached, we would find that the product of L (in millihenrys) multiplied by C (in microfarads) would be 25. At some other frequency, we would find the product of L times C would be equal to some other number; in other words, there is one number corresponding to each resonant frequency. Tables are available giving these numbers for different frequencies, so an engineer who is assembling a resonant circuit need only look up this number for the frequency in question, then choose values of L and C which, when multiplied together, will give this number. (Although this number is equal to L times C, engineers for convenience often omit the word "times" and simply call it the L-Cvalue.)

This relationship between L and Cat any resonant frequency makes it easy for engineers to predict beforehand what values of L, C, and frequency will give resonance in a particular circuit. When any two of these values are known, they can find the third either by means of a chart, a table, or a mathematical formula.* We just said that the product of L and C for a circuit resonant at 500 cycles is 100, while L times C for a circuit resonant at 1000 cycles is 25. This is an example of a general rule which you will find useful to remember—the lower the L-C value of a series resonant circuit, the higher the frequency to which the circuit is resonant. In other words, decreasing the value of either L or C increases the



FIG. 6. Here is what happens when you vary the frequency of the source in a series resonant circuit. These curves prove that for any given set of values for L and C, there is only one frequency at which resonance and maximum current will be obtained. Notice that it is possible to obtain resonance with a different value of L and C, providing the L-C product is the same. Also, reducing the circuit resistance (curve B) pro-

duces a sharper curve with a higher peak.

resonant frequency of a series resonant circuit, while *increasing* the value of L or C decreases the resonant frequency.

* For those who are interested in mathematics, this formula is:

 $\mathbf{f} = 159,000 \div \sqrt{\mathrm{LC}}$

In this formula, f is the resonant frequency in cycles, I, is the inductance of the coil in microhenrys (1 millihenry = 1000 microhenrys) and C is the condenser capacity in microfarads. This formula applies to both series and parallel resonant circuits. For example, using a 100-mh. coil and a condenser of 2 mfd. does not give resonance at 500 cycles, but does give resonance at the frequency corresponding to an L-C value of 200. As we expect, the new resonant frequency is lower than 500 cycles—it is near 355 cycles. Hence, when we change either the coil or the condenser, we tune away from one frequency but establish resonance at another. Thus, an L-C circuit is always resonant to some frequency, although it may not be the desired frequency.

Resonance Curves. The most important characteristic of a resonant circuit is how much better it accepts the desired resonant frequency than the undesired frequencies on either side. In radio receivers, for example, we want the resonant circuits to tune in a desired carrier frequency signal while rejecting the adjacent undesired carriers. To find how well a resonant circuit can do this, we refer to a curve obtained by plotting current against source frequency. (Such a curve is commonly called a resonance curve, because it shows how circuit conditions change as we pass through resonance.) Three such curves for series resonant circuits are shown in Fig. 6: let us see how much they can tell.

 \blacktriangleright Curve A in Fig. 6 is for the series resonant circuit in Fig. 2; it was obtained by keeping the source voltage constant while varying the frequency and measuring the circuit current. You can see that there is quite a drop in circuit current as you go above and below the resonant frequency of 500 cycles. The sharper the curve at resonance, the greater will be this drop in current or, as radio men say, the better will be the selectivity of the circuit. (Since such a curve shows how selective a circuit is, it is often called a *selectivity* curve.) You will learn later in the course, however, that a certain amount of broadness in the curve at resonance is also desirable in order to keep distortion at a minimum. Let us see how we can get a more peaked curve, and a broader curve.

 \blacktriangleright You will recall that the rheostat (R) was set at 120 ohms in Fig. 2; if we change the setting to 100 ohms, the circuit resistance at resonance will be lower and we will secure a more peaked resonance curve. This is evident in Fig. 6 from the fact that curve B for 100 ohms has a higher and sharper peak than curve A for 120 ohms. Lowering the resistance in a series resonant circuit thus improves selectivity, as it allows a higher cur-This means the rent at resonance. resonant step-up is greater, but the voltages at other frequencies are not stepped up, so the difference between the voltage at resonance and the offresonance values is greater.

 \blacktriangleright Suppose we use a 50-mh. coil and a 2-mfd. condenser, resetting R to 120 ohms. We still get resonance at 500 cycles, as the L-C value of 100 has not been changed, but now curve C in Fig. 6 represents conditions in the circuit. Note that this curve is considerably broader at resonance; this indicates that a wider range of frequencies around the resonant value will be passed almost equally well. If we experimented further in this same way, we would find that the lower the value of L, while circuit resistance is kept constant, the broader will be the resonance curve obtained. (Incidentally, resonance or selectivity curves like those in Fig. 6 are also known as frequency response curves or simply as response curves.)

Resonance Current and Voltage. We have pointed out several times that a circuit will be resonant at 500 cycles regardless of what values of Land C we use, provided only that the product of L times C equals 100. The circuit current will be about the same at resonance whether we use a large coil and a small condenser or a smaller coil and a larger condenser. This is because the reactances of L and Ccancel at resonance, no matter what their values are, and only the resistance in the circuit limits the current flow.

The voltages across the coil or condenser at resonance, however, will depend upon the values of L and C. If we obtain resonance by using a 50-mh. coil, for instance, the voltage across it will be only about half what it would be across a 100-mh. coil. You can readily see the reason for thisthe voltage across the coil is equal to the current times the coil reactance. A 50-mh, coil has only half the reactance of a 100-mh. coil and, since the current remains constant, reducing the coil reactance one-half reduces the voltage across it by one-half. For this reason, in series resonant circuits a high L-to-C ratio is used when the greatest resonance voltage step-up is wanted. (A large inductance and a small capacity are used.) Don't confuse this ratio with the L-C value; the ratio is $L \div C$, while the L-C value is $L \times C$.

Parallel Resonant Circuits

When the source for a resonant circuit is connected in parallel with both the coil and the condenser, as in-Fig. 7, we have a parallel resonant circuit. The values of L, C, and R used in this circuit are the same as those in the series resonant circuit in Fig. 2, so you might expect that circuit current and voltages would be about the same for Fig. 7 as they are for Fig. 2. But actual measurements would show that they are completely different.

For one thing, the circuit current in Fig. 7 (measured on an a.c. ammeter in series with the generator) is very low. (As you recall, the circuit current in a series resonant circuit is high.) And, practically all the generator voltage appears across the resonant circuit, between terminals X and Y, with almost no voltage across resistance R. This is completely different from the series resonant circuit.



FIG. 7. At resonance, a parallel resonant circuit reduces the line current to a very low value.

In the series circuit, nearly all the generator voltage appears across resistance R, and almost none across the resonant circuit. (Of course, there is always a high voltage across each of the *parts* in the series circuit, but these voltages cancel one another. Across the *whole* L-C circuit, between terminals X and Y in Fig. 2, there is almost no voltage.)

Suppose we set up a circuit like that in Fig. 8, with a.c. ammeters inserted in the parallel resonant circuit to measure currents I_1 -and I_2 through C and L, respectively. We would discover another surprising fact—there is plenty of current inside the parallel circuit itself! I_1 and I_2 are both high, and equal to one another. Yet, outside the parallel circuit, there is practically no current.

Clearly, this resonant circuit for some reason acts like a very high resistance across its terminals (X andY in Fig. 8), but at the same time offers very little opposition to current flow in the resonant circuit itself (the circuit made up of L, C, and the two ammeters). The reason for this peculiar combination of actions is really quite simple. Here, our reference factor is the source voltage which is across both the coil and the condenser. You know that the current flow in a coil is 90° behind the coil voltage, while the condenser current is 90° ahead of the condenser voltage. Since the source voltage is now across both the coil and the condenser, it becomes the coil voltage and the condenser voltage. As one current is ahead 90° and the other is behind 90°, this makes the currents 180° out of phase. (Compare this with the series resonant circuit, where the *current* is the reference value and the *voltages* are 180° out of phase.)

The fact that the currents I_1 and I_2 are 180° out of phase means that the condenser discharges during the half cycle in which the coil stores up energy; on the next half cycle, the coil releases energy while the condenser stores it. When the impedances of the coil and the condenser are equal (that is, at resonance) the energy stored by the condenser equals the energy released by the coil, and vice versa. Then the condenser absorbs all the energy released by the coil, and in turn releases this energy to be absorbed by the coil. Thus the coil and condenser pass current back and forth



FIG. 8. The current within a parallel resonant circuit is very high, compared to the line current, due to resonance current step-up.

inside the resonant circuit, causing the high readings on the meters in the circuit. The source has to supply only enough power to make up for the small power losses in the small resistances within the coil and the condenser, so the current outside the resonant circuit is small.

► Whenever a coil and a condenser are in parallel in an a.c. circuit (regardless of whether or not their reactances are equal), one will be feeding current back into the line while the other is drawing current. Therefore, the line current at any instant (taking phase into account) will be the difference between the two currents. When the reactances are equal, as at resonance, the line current is a minimum, while both the coil and condenser draw a current determined by their reactances and the voltage across the resonant circuit. At resonance, then, a parallel resonant circuit behaves as a resistor of high ohmic value, reducing line current almost to The resistance of a parallel zero. resonant circuit at resonance is known as the resonant resistance of the circuit.

Resonant Current Step-Up. You have just seen that the current through either the coil or the condenser in a parallel resonant circuit is higher than the line current at resonance. This is *resonant current step-up*—the ability of a parallel resonant circuit to circulate a current many times greater than the current fed into it. This particular characteristic is very useful in radio circuits, as you will soon see.

VARYING R, L, C, AND F

Varying the Coil Resistance. Since all coils have a certain amount of resistance acting in series with their inductance, let us see how increases in this resistance affect a parallel resonant circuit. Instead of analyzing circuits using coils with the same inductance and various values of resistance, let us assume a circuit like that in Fig. 9, where we have a rheostat R_1 in series with our coil. If we were to start with R_1 at zero resistance, then increase it gradually, we would find that the coil current goes down only a little, the condenser current remains the same, but the line current goes up. This means that when the resistance in series with the coil in a parallel resonant circuit is increased. the resonant resistance of the L-C cir-



FIG. 9. Resistance within the resonant circuit (R1) reduces the current step-up.

cuit (between points x and y in Fig. 8) is decreased. This allows more line current to flow, and makes the line current value more nearly equal to the condenser current. We can conclude, then, that inserting resistance in a parallel resonant circuit decreases both the resonant current step-up ratio (the ratio of coil current to line current) and the resonant resistance.

Varying C. With the parallel resonant circuit in Fig. 9 tuned to resonance and both R and R_1 at their zero resistance settings, let us see what effect varying C would have on the line current while the source frequency is kept constant at 500 cycles. Fig. 10, in which the line current is plotted for various values of capacity, shows the result we would get.

Starting with a very small value of capacity for C, we would find that condenser current is practically zero, as it offers a very high reactance. The line current is then determined by the coil reactance (R is set at zero). Hence, the coil and line currents are both about .35 ampere.

As the capacity of C is gradually increased, its reactance decreases, so we would find that the condenser current goes up, line current goes down, and coil current remains unchanged. Coil and condenser currents become more nearly equal as the capacity of C approaches its resonant value. Finally, the condition of resonance is reached, where line current is at its minimum value and the coil and condenser currents are equal.

Increasing the capacity of C above the resonant value causes line current to go up again, while condenser current continues to increase. With a very high capacity for C the reactance of C becomes so low that coil current becomes negligible in comparison with the extremely high condenser current, and meters I and I_1 of Fig.



FIG. 10. This graph tells how the line current changes when the capacity of the condenser (C) in a parallel resonant circuit is varied.

9 read almost the same values.

 \blacktriangleright Lowering the capacity of C below the resonant value (the value giving resonance at the source frequency) makes the reactance of the condenser greater than the reactance of the coil. A parallel resonant circuit acts the same as the part drawing the most current (having the least reactance), so in this case the parallel circuit acts like a coil. On the other hand, raising the capacity of C above the resonant value makes the reactance of the condenser less than that of the coil, so the parallel circuit acts like a condenser.

► Actually, when we reduce or increase the value of C from the resonant value, we make the parallel circuit resonant to some other frequency. The frequency to which the circuit will be resonant with the changed value of C is, of course, the frequency at which the reactance of the coil will equal the reactance of the new value of C. From what you already know about how the reactances of coils and condensers vary with frequency, it will be easy to see that decreasing the value of C raises the frequency to which the circuit is resonant, while increasing the value of C lowers the resonant frequency.

Varying L. If we were to vary inductive reactance (instead of capacitive reactance) by using different sizes of coils for L while keeping the source frequency constant, we would find that low-inductance values of L had the same effect as high-capacity values of C, and high values of L gave the same results as low values of C.

Varying Frequency. Now suppose we see what would happen in the circuit shown in Fig. 9 if we were to set both R and R, to zero resistance and vary the frequency of the source without changing its voltage or the coil and condenser values. Suppose we start with a very low frequency (less than one cycle per second). This is very nearly the equivalent of direct current, and you know that C will act as an open circuit (very high reactance), while L will act as a short (very low reactance). This means that a very high current will flow through the coil and ammeters Iand L.

Increasing the frequency gradually from zero to 500 cycles makes the reactances of L and C more and more nearly equal, with the parallel resonant circuit acting in the same manner as a coil (because below 500 cycles



FIG. II. The effects of variations in frequency on the line current for a parallel resonant circuit are given on this graph for two different L-C ratios.

the reactance of L is lower than that of C). Line current drops to almost zero at 500 cycles, and condenser current increases gradually, being equal to the coil current at 500 cycles.

▶ When the source frequency is 500 cycles, we have the resonance condition already described.

As the source frequency is increased above 500 cycles, coil reactance goes up and condenser reactance goes down, making our parallel resonant circuit behave as a condenser, with line current increasing again to supply the extra current drawn by the condenser. At a very high frequency, the reactance of L will be so high that it will act as an open circuit, but the reactance of C will be practically zero, and line current will be just as high as it was for very low frequencies.

Varying Both L and C. If we were to insert various values of L in Fig. 9 and use the correct values of C to give resonance at 500 cycles, we would find in each case that the same number is obtained by multiplying together the values of L and C. In other words, the product L-C is a constant for a given frequency, just as it was for series resonant circuits. Furthermore it is the same number as the L-C value for series resonance. Also, the formula for determining the resonant frequency when L and C are known (already given you in a previous foot note) applies to parallel resonant circuits as well as series resonant circuits.

Resonance Curves. We can draw the same kind of resonance curves for our parallel resonant circuit that we draw for the series circuits by plotting frequency against line current, as in Fig. 11—the entire 110 volts is applied, as R is reduced to zero. When L is 100 mh. and C is 1 mfd., curve Ais secured, while when L is 10 mh. and C is 10 mfd., curve B is the result. As you can see from these curves, lower values of L in parallel resonant circuits make line current increase more rapidly off resonance. In other words, a parallel resonant circuit tunes more sharply if L is small, so a low L-to-C ratio is desirable here. A small inductance and a large condenser are used for greatest selectivity.

Practical Resonant Circuit Facts

Tuned circuits are found most commonly in the r.f. stages of radio apparatus, where they are used to separate signals coming from different stations. This separation is possible when the stations operate on different frequencies, as we will show you later in this lesson by means of typical circuits.

However, there are several points we should clear up before going on to these practical circuits. We should get an idea of the amount of resonance step-up obtainable; learn how to recognize series and parallel circuits; and learn how these circuits may be adjusted or tuned to resonate with different frequencies. We will now take these subjects up in order.

Q FACTOR

You have learned that the resonant voltage step-up or resonant current step-up of any resonant circuit increases as the resistance in the circuit is decreased. Since a high step-up ratio usually is wanted, engineers seldom insert resistors intentionally in resonant circuits. However, any resonant circuit has a certain amount of loss caused by the effective a.c. resistances of the coil, the condenser; and the wiring in the circuit. Usually we can assume this loss is concentrated in the coil. And we can see what effect such a loss has on a circuit by using a quantity called the "Q factor" of the coil.

The Q factor is really a measure of

the merit of the coil. It can be computed for any coil from the formula $Q = X_L \div R$, where R is the total *a.c.* resistance of the coil, and X_L is the inductive reactance of the coil. (Test instruments are available which will measure the Q of a coil.) As practically all the loss in a resonant circuit is concentrated in the coil, we usually assume that the coil Q is also the Q of a resonant circuit using that coil. This factor applies to both series and parallel resonant circuits.

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► Of what use is the Q factor to us in our study of resonant circuits? For one thing, it allows us to compare two coils of the same inductance to see which has the more desirable characteristics for use in a tuned circuit. If two coils have the same inductance, but the Q factor of one is higher than the other, we know at once that the coil with the higher Q will give sharper tuning (better selectivity) in a resonant circuit. (This follows from what you have already learned. You know that a circuit tunes more sharply if the resistance in it is small, and from the formula just given, it is obvious that if two coils have the same inductance, the coil with the higher Q factor will have the smaller resistance.)

Also, a knowledge of the Q factor of the coil used in a resonant circuit will let us find out several things about the circuit itself. In a series resonant circuit, the applied voltage multiplied by the Q gives the voltage available across either the coil or the condenser. In a parallel resonant circuit, the line current multiplied by Q gives the current flowing in the tuned circuit; also, multiplying Q by the coil reactance will give us the resonant resistance of the parallel resonant circuit.

 \blacktriangleright Notice—we used the *a.c.* resistance of the coil in our formula for Q factor. This is not the same amount as the d.c. resistance of the coil, which we could measure with an ohmmeter. The a.c. resistance of a coil is always much higher than the d.c. resistance, particularly when the coil is used in r.f. circuits. This is because r.f. current does not travel through the entire cross-section of the wire in a coil, but instead tends to travel close to the wire surface. The tendency becomes more marked as the frequency of the current is increased. . This is known as the "skin effect."

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The skin effect means that the actual current-carrying area of the wire in a coil is considerably reduced as the frequency of the current increases, and therefore the resistance of the coil increases. To this resistance must be added dielectric losses in the coil form, shield losses, and, in a coil which has an iron core, losses in the core. All these factors make up the a.c. resistance of the coil as they represent losses. This resistance has all the characteristics of resistance, and must not be confused with reactance, which the coil also has. Engineers frequently draw a resistance symbol in series with the coil symbol to indicate that the coil resistance is in series with the inductance.

Since both the a.c. resistance and the inductive reactance increase as the frequency of the circuit current increases, it might seem as if these increases would cancel one another, so that the Q factor of the coil (the ratio of the reactance to the a.c. resistance) would remain constant for all freauencies. In the interest of greater operating efficiency, many coil designers try to achieve this effect. Usually, however, the a.c. resistance of a coil increases much more rapidly than does the coil reactance as the frequency of the circuit current is raised. Therefore, it is generally true that the Q factor of a coil becomes smaller when higher frequency currents flow through it. In recognition of this fact, the Q factor of a coil always is specified as having been measured at some particular frequency.

COMPARING RESONANT CIRCUITS

A comparison of series and parallel resonant circuits is given elsewhere in this lesson, in a table. Here are a few additional facts:

► When a resonant circuit is designed for a particular frequency, inductance and capacity values are chosen which will give the $L \times C$ value necessary at that frequency. However, we often want a resonant circuit which can be tuned over a band of frequencies at will. For example, the r.f. circuits of your radio tune over the broadcast band, covering frequencies from 550 kc. to 1500 or 1600 kc. In these cases, the maximum inductance or capacity value of the variable part is found, then the circuit is designed to tune to the lowest frequency in the band. Then, as the variable capacity or variable inductance value is decreased, the circuit tunes to higher frequencies and thus is made to cover the desired band. In this case, notice that there are a number of frequencies applied simultaneously, and the resonant circuit is adjusted to "tune in" or accept the one in which we are interested.

► When a choice of several L and C values is possible and maximum selec-



tivity is desired:

- A. For series resonance, use a large inductance and small capacity, giving a high $L \div C$ ratio.
- B. For parallel resonance, use a small inductance and large capacity, giving a low $L \div C$ ratio.

• After the L and C values have been chosen, the Q factor then determines the actual voltage or current step-up. The higher the Q (the lower the a.c. resistance), the greater the step-up.

Distinguishing Between Series and Parallel Resonant Circuits. One question often asked even by experienced radio men is, "How can I tell whether a resonant circuit is of the series or parallel type when I see it in a circuit diagram or in radio apparatus?" It's not really very difficult. Here's a simple procedure that will give you the right answer every time: 1. Locate the coil and the conden-

ser forming the resonant circuit.

2. Locate the source of a.c. voltage for this resonant circuit. Typical sources will be receiving antennas, plate circuits of vacuum tubes, and voltages induced from other circuits.

3. If the source for the resonant circuit is *in series* with the coil and the condenser, you have a series resonant circuit.

4. If the source for the resonant circuit is connected *directly in parallel* with both the coil and the condenser, you have a parallel resonant circuit.

VARIABLE INDUCTANCES

You know that a circuit can be tuned either by varying either the capacity of the condenser or the inductance of the coil, but how do we actually vary these things? You know the answer as far as variable condensers are concerned, since they have been fully covered in another lesson. Variable inductances were mentioned in a previous lesson, but now let's spend a little time learning more about how variable inductances work, because inductance tuning is becoming increasingly more important in modern radio receivers. (It has been used in transmitters for years.)

You will recall that anything which increases or decreases the magnetic flux linkage per ampere of current flowing through a coil will increase or decrease the inductance of the coil. For example, increasing the number of turns on the coil increases its inductance, and increasing the amount of flux flowing through the coil like-



FIG. 12. Methods of varying the inductance of a coil.

wise increases its inductance. One way of building a variable inductance, then, is to use a coil which has the number of turns required to give the maximum desired inductance for a circuit, and provide some means of using fewer turns when we want to reduce the inductance in the circuit. One scheme which has been widely used for this purpose is shown in Fig. 12A. Here, various turns on the coil are connected by means of wire leads (taps) to metal contacts over which a contact arm moves, thus permitting the operator to select as many or as few turns as he requires.

Earlier radio receivers used this tapped inductance extensively, while a variation of it is still being used in transmitters. Fig. 12B shows a transmitting coil with movable clamp type contacts which can be placed on any desired turns. Only those turns which carry current are effective in producing flux linkages.

▶ You can increase flux through a coil and thereby increase its inductance by placing inside the coil a material which offers less opposition than air to the flow of magnetic lines of force.

For low frequencies, iron and steel in either solid or laminated (thin sheets in layers) form make satisfactory coil cores, but for radio frequencies, solid or laminated cores act as a multitude of tiny short-circuited secondary coils which serve to reduce Radio the useful magnetic flux. manufacturers get around this difficulty by pulverizing the steel core material and mixing this steel powder with liquid bakelite (a good insula-The resulting mixture can be tor). molded into a core of the desired shape for use with radio frequency coils. The metal particles are insulated from each other by the bakelite and so do not produce the short-circuiting effect just mentioned. However, enough iron is in the mixture to reduce the reluctance of the magnetic field and thus to increase the inductance. If the core is designed so that it can be moved in and out of the coil, as indicated in Fig. 12C, we have another form of variable inductance. This style of inductance variation is used most commonly today in radio receivers.

► Two coils are often connected together in series. If the coils are some distance apart, so that there is no coupling between them, their combined inductance is the sum of the separate inductances. As the coils are moved closer to each other, so that the flux of one links the turns of the other, the combined inductance will increase if the two fluxes are in the same direction and will decrease if the two fluxes are in opposite directions. A variable inductance of this type is illustrated in Fig. 12D.

A more convenient way of using the combined inductance of two coils to produce a variable inductance is shown in Fig. 12E. Here, one of the coils is made smaller and mounted on a shaft inside the other. It is possible, by rotating the inner coil, to vary the combined inductance of these coils over a wide range. When the coils are arranged so that their currents flow in the same direction, the fluxes will be in the same direction and the combined inductance will be high. When the inner coil is turned halfway around (180°) so that the currents flow in opposite directions, the fluxes will oppose each other and the inductance will be greatly reduced. A variable inductance of this type is known as a variometer.

► Another method of constructing a variable inductance is illustrated in Fig. 12F. A rotatable metal ring is mounted in the center of the core; when this ring is at right angles to the axis of the coil, it acts as a shortcircuited secondary winding having one turn. The induced voltage causes a high current in this secondary, which in turn produces a flux which partially balances out the coil flux, reducing the inductance of the coil. Rotating the ring lessens the amount of flux which can induce a voltage in the ring. This reduces the "bucking" effect of the flux produced by the ring and increases the coil inductance. A metal disc will work equally as well as a

ring in this variable inductance, for a disc really consists of a great many rings, one inside the other. A very important fact to remember is that the presence of any conducting material near a coil, such as a metal shield surrounding a coil, will reduce the inductance.

 \blacktriangleright Of course, a fixed condenser is used in the tuned circuit when the inductance is variable, while a fixed inductance is used with a variable condenser for any one tuning range. Where several ranges are to be covered, as in a receiver designed to tune over short-wave bands as well as the broadcast band, tapped coils (or separate coils, exchanged by a switch) usually are used for the extra bands. You will learn more about this when you study r.f. stages and all-wave receivers.

How Resonant Circuits Are Used

Now, since you have learned how resonant circuits function, let us put both the series and the parallel circuits to use in practical radio applications.

WAVE TRAPS

Using a Parallel Resonant Circuit. When a radio receiver is located so near a powerful short-wave or broadcast band transmitter that signals from this station interfere with those of stations broadcasting on adjacent frequencies, a wave trap similar to that shown in Fig. 13A is quite often used. The wave trap weakens the signal of the undesired local station before this signal gets into the amplifying section of the receiver. parallel resonant circuit made up of coil L and condenser C is inserted in the antenna lead-in wire and tuned to resonance at the frequency of the undesired signal.

Fig. 13B shows schematically how this wave trap works. The antenna and ground system can be considered a voltage source, while the radio itself acts as a load for this source. Our wave trap is inserted in the circuit between the source and the load. Now, the source voltage will always divide between the wave trap and the load

in proportion to their impedances. When the source voltage (that is, the input signal) has the frequency to which the wave trap is tuned, the wave trap has a very high impedance, while the load (the radio) has its moderately low impedance. usual Therefore, nearly all of this particular signal voltage is developed across the wave trap, and very little is fed to the radio. At other frequencies, to which the wave trap is not tuned, the trap will have much lower impedance than the radio. Then most of the signal voltage will be developed across the radio, and very little will be wasted in the trap. Thus, our wave trap acts as a signal rejector; it keeps one narrow band of frequencics out of the radio, but lets all frequencies above and below this band go on to the set. This circuit is actually a filter. Since it eliminates a narrow band of frequencies and lets those above and below pass, it is called a band-elimination filter.

A coil and a condenser connected to form a parallel resonant circuit which can be used as a wave trap are shown in Fig. 13C. The trimmer type condenser need be adjusted only once, at the time when the wave trap is installed.



FIG. 13. Using a parallel resonant circuit as a wave trap.

Using a Series Resonant Circuit. A series resonant circuit, when connected to the antenna and ground terminals of a receiver in the manner shown in Fig. 14A, also serves as a wave trap. We can analyze the action of this trap with the schematic diagram in Fig. 14B. Again, the antennaground system is represented as a voltage source and the receiver is shown as a load. The series resonant circuit is across this load. Notice-we have included an impedance Z in the voltage source. This is the impedance of the antenna-ground system. (This impedance was present also in the circuit we just discussed, but was not mentioned because there it was unimportant in the analysis of that circuit.)

Since the load and the series resonant trap are in parallel, there always will be the same voltage drop across each. The source voltage will divide between the impedance Z of the source and the combined impedances of the trap and the receiver. At the resonant frequency of the trap, its impedance will be very low-and, of course, the combined impedances of the trap and the receiver will be even lower. Thus, when the input signal is of the frequency to which the trap is tuned. practically all the voltage will be dropped across impedance Z, and very little will be dropped across the trap and the receiver. At other frequencies, the trap-receiver combination will have much higher impedance than Z, and nearly all the input signal voltage will be dropped across them. Again, we have a signal rejector which keeps one narrow band of frequencies out of the radio receiver.

The actual connections of a coil and condenser used as a wave trap of the series resonant type are shown in Fig. 14C. A completely assembled wave trap in a neat metal housing is illustrated in Fig. 14D. This is of the series resonant type, with connections like that in Fig. 14A; hence its two terminals are connected directly to the antenna and ground terminals of the receiver. The condenser can be adjusted by inserting a screwdriver through a hole in the top of the housing, and turning the adjusting screw.



FIG. 14. How a series resonant wave trap should be connected.

²⁰ World Radio History



FIG. 15. A parallel resonant circuit used as a plate load in a vacuum tube circuit. A variable condenser may be used instead of the trimmer shown at C.

▶ Be sure to remember two very important facts about the two types of wave traps. A parallel resonant trap is connected in series with the receiver. A series resonant trap, on the other hand, is connected in parallel with the receiver. (A parallel resonant trap is often called a series trap, and a series resonant trap is often called a series trap, and a series resonant trap is often called an parallel trap, because of the way they are connected into the circuit. These terms, however, refer only to the method of connection, not to the kind of resonant circuit used.)

Wave Trap Data. Wave traps of the parallel resonant type generally give best results on receivers which have a low input impedance, while those of the series resonant type will work best with receivers having a high input impedance. When the impedance of the receiver input is unknown, the practical radio man usually tries both types, selecting the one which gives the greater interference reduction.

Adjusting a wave trap is a simple matter. Connect it to the receiver in the proper manner, tune the receiver to the offending station, and then adjust the condenser in the wave trap until the offending signal is as weak as possible. If more than one station interferes, use one wave trap for each interfering station. Many all-wave superheterodyne receivers have builtin wave traps, which can be adjusted by the serviceman to tune out offending low-frequency code stations.

A wave trap is not a cure-all; there are many types of interference which it cannot reduce. You will learn more about the types of interference in later lessons.

TUNING CIRCUITS

Parallel Resonant Circuits in R.F. Amplifiers. Tuned or resonant circuits are widely used for feeding radio frequency signals from one tube to another, as in tuned r.f. amplifiers. The tuned-impedance or high-gain amplifier circuit shown in Fig. 15A is an example. We are interested primarily in the action of the L-C circuit drawn in heavy lines, since the other parts of this circuit will be taken up in detail later in the Course.

For our purpose, we can represent the input side of this circuit (tube VT_1 and the *L-C* circuit) by the schematic diagram in Fig. 15B. Here we consider tube VT_1 to be an a.c. voltage source with a resistance R_P . You can see at once that any voltage produced by the tube source will divide between the resistance R_P and the parallel resonant circuit.

If we tune the parallel circuit to resonance at the frequency of the incoming signal, it will have a very high resistance at this frequency. Consequently, most of the signal voltage of tube VT_1 will be developed across the parallel circuit and will be passed on to the grid of tube VT_2 through $C_{\rm K}$. The parallel resonant circuit will have a low impedance for input signals of other frequencies, however, so they will be largely dropped in $R_{\rm P}$ and little energy will be passed on.

A parallel resonant circuit connected this way passes on signal voltages at the frequency to which it is tuned, and rejects voltages at all other frequencies. This means that the circuit possesses selectivity characteristics which make it very valuable for separating r.f. signals, so we can "pick out" and accept those from a desired station.

The picture diagram in Fig. 15C shows you the actual connections for a parallel resonant circuit, just as you might find them in the chassis of an r.f. amplifier. Notice how much simpler it is to draw the schematic diagram (Fig. 15A) than the picture diagram. Except for keeping certain leads (such as grid and plate leads) as far apart as possible, the manner in which parts are physically connected together is generally not important as long as the proper terminals are connected together *electrically*.

Series Resonant Circuits in R.F. Amplifiers. Another typical r.f. amplifier which uses a resonant circuit is shown in Fig. 16A. There is no difficulty here in recognizing the resonant circuit made up of condenser Cand coil L, but the question is: Where is the source of voltage for this circuit? We must locate this source before we can tell whether the circuit





is of the series or the parallel type.

The answer to this question lies in the transformer made up of coils L_P and L. The r.f. signal current supplied by the plate of tube VT_1 flows through coil L_P and induces an r.f. voltage of corresponding wave form in coil L. The field about L_P links with L, inducing a voltage WITHIN L, just the same as if a voltage source were placed in series with L. In fact, we can represent the resonant circuit by the diagram in Fig. 16B, where the voltage induced within L is shown as the generator E.

By tracing from the generator in Fig. 16B, you can see that to make a complete circuit we must go through part of coil L, through condenser C, and through the other part of coil L back to the generator. Therefore, this is obviously a series circuit—as is any resonant circuit in which the voltage is induced in the coil.

The identification of a circuit of this kind puzzles many radio men. Remember this one fact, and you'll have no difficulty in identifying the circuit:

Whenever a voltage is induced in the coil of a tuned circuit, the circuit is series resonant.

When L and C in this series resonant circuit are tuned to the frequency of the induced signal voltage, they act as a resistance of very low ohmic value. A large r.f. current, therefore. flows through the series resonant circuit, developing a correspondingly large voltage across condenser C. and this voltage acts on the grid and cathode of tube VT_{a} . At all other frequencies, the L-C circuit acts as a high reactance. The r.f. currents at these other frequencies are therefore low, so the voltages developed across C are low, thus the desired signals are favored. A series resonant circuit thus gives selectivity besides giving resonant amplification of the desired signal. Chassis connections as they might be for a series resonant circuit in an r.f. amplifier are shown in Fig. 16C.

Resonant Circuits in a Superheterodyne Receiver. You have just seen how series and parallel resonant circuits each contribute a certain amount of selectivity. In superheterodyne receivers, series and parallel resonant circuits are often connected together, in the manner shown in Fig. 17A, to secure even greater selectivity. Again the resonant circuits are shown in heavy lines.



FIG. 17. In the i.f. section of a superheterodyne, both series and parallel resonant circuits are used.

First consider only circuit L_1 - C_1 ; clearly, this is a parallel resonant circuit as is that shown in Fig. 15A, because tube VT_1 is its signal source. At resonance, then, a large current flows through coil L_1 and condenser C_1 .

Now let us bring the L_2 - C_2 circuit into our picture. The r.f. current flowing through coil L_1 induces in coil L_2 an r.f. voltage which acts in series with L_2 and C_2 , just as in Fig. 16A. Thus L_2 and C_2 form a series resonant circuit. The actual chassis connections for this double resonant circuit might be as shown in Fig. 17B.

Coupling Radio Circuits

Now that we are well acquainted with resonant circuits and with a few of their applications, we are ready to take up the different ways in which a resonant circuit can be coupled (connected) to a source of voltage.

ANTENNA COUPLING METHODS

The antenna circuit of a radio receiver is one of the simplest and bestknown applications for coupling systems, so we will use it as our practical example in connection with this study of coupling methods. Suppose we have an antenna and a ground,



FIG. 18. How can we couple the resonant circuit to the antenna-ground?

and want to connect or couple them to the resonant or tuned circuit which feeds the first r.f. stage of a radio receiver, as shown in Fig. 18A.

The antenna-ground circuit will be a source of voltage, as it picks up energy from the traveling radio waves. However, like all other sources, the antenna-ground circuit has an impedance. It has a certain amount of resistance and also has some inductive and some capacitive reactance. Hence, the antenna-ground circuit can be represented as in Fig. 18B. As you can see, it is a series resonant circuit itself, but in practice antennas are so dimensioned that they are resonant only in the short-wave bands. Thus, at the lower or broadcast frequencies, the combination is far off resonance, and acts like a condenser in series with a resistance.

We can lump together all the opposition as an impedance, as shown in Fig. 18C. We don't need to know the exact amount of either the impedance of Z_A or the voltage of E, but we do want to know how we can couple from this circuit into the resonant circuit.

There are four simple ways to couple an antenna to a single resonant circuit:

- 1. Direct coupling.
- 2. Capacitive coupling.
- 3. Direct inductive coupling.
- 4. Inductive coupling.
- Let us consider them one by one.

1. Direct coupling, as shown in Fig. 19A, is the first and simplest solution to this problem. The entire antenna voltage is applied across our L-C circuit, making it a parallel resonant circuit. The resonant circuit then becomes the load, and forms a voltage divider with the antenna impedance Z_A , as shown in Fig. 19B.

As you have learned, this tuned circuit will act as a resistance of high ohmic value to a signal having the resonant frequency to which L and C



FIG. 19. Direct coupling.

are tuned. At this desired frequency, then, the resistance of the input circuit is much higher than the antenna impedance, and the greatest part of the signal voltage will exist across the resonant circuit and be passed on to the grid of the first tube. At all other frequencies, however, this resonant



FIG. 20. Capacitive coupling.

circuit will act as a low impedance, and a far smaller voltage will be developed across it.

2. Capacitive coupling, as illustrated in Fig. 20A, is the second method. As you can see, it is simply direct coupling with a low-capacity condenser $C_{\rm K}$ inserted in the antenna lead-in.

Using the condenser gives an improvement over the direct coupling circuit shown in Fig. 19. In direct coupling, the selectivity is more sharp the higher Z_A becomes. The voltage division for off-resonant signals is not effective until the resonant circuit impedance has dropped to some value near (or below) the Z_A value. Therefore, when Z_A is small, the parallel resonant impedance must drop a great amount before off-resonant signals are eliminated. This means signals near the resonant frequency are not cut out, so the tuning is broad.

Inserting the condenser $C_{\rm K}$ as shown in Fig. 20A places its reactance in series with the antenna impedance as shown in Fig. 20B. Hence, it is added to the series impedance, increasing the effectiveness of the circuit. Now the parallel impedance does not have to drop so far before its value compares with that of the series impedance, so the circuit rejects frequencies closer to the resonant value.

3. Direct inductive coupling of the form shown in Fig. 21A applies the antenna voltage to only a part of the resonant circuit. By having the tap 3 well up on the coil (near 1), the circuit still acts as a parallel resonant circuit, but only a part of its impedance will be between 2 and 3, and hence in the antenna-ground circuif. This can result in a better voltage-dividing action, and transformer step-up can be used to increase the voltage slightly. The position of point 3 can be chosen to secure maximum gain with fair selectivity or low gain with good selectivity, as desired.

4. Inductive coupling, shown in Fig. 22A, is perhaps the most widely used of all methods for coupling the antenna system to a resonant input circuit. (Inductive coupling is also known as transformer coupling or sometimes as indirect inductive coupling.) Antenna current flowing through



FIG. 21. Direct-inductive coupling.

coil L_1 sets up a varying magnetic flux M which links coil L and induces in it a voltage which acts as if it were in series with L. Inductive coupling can be designed to give very high gain, to give high selectivity, or to give a satisfactory compromise between the two, as desired, by varying the num-



FIG. 22. Inductive coupling.

ber of turns in coil L_1 and its position with respect to coil L.

The coupling methods shown in Figs. 19 to 22 for antenna circuits are essentially the same as those you will encounter in a host of other radio applications, the most common of which is the coupling together of the plate circuit of one tube and the grid circuit of the following tube. We will study these circuits in detail in later lessons, when we take up complete radio stages.

INDUCTIVE COUPLING FACTS

Any inductively coupled circuit can be simplified to the three essential parts shown in Fig. 23: a source, a coupling transformer, and a load. The source may or may not have appreciable impedance, and may be delivering either an a.f. or an r.f. signal voltage. The transformer may be of the air-core type, if its two windings (coils) are separated only by air or some other insulating material, or it may be of the iron-core type. The



FIG. 23. Any inductively coupled circuit can be represented by this simplified circuit.

load likewise may have impedance. Either the source, the load, or both, may be resonant circuits. In an inductively coupled circuit of this sort, the source sends through the transformer primary (coil L_1) a current I_P which induces in coil L_2 a voltage that acts directly on the load, making current I_s flow through the load.

The current flow through the load means that the load is taking power from the coupling transformer, and further, this power must be given to the coupling transformer by the source. Hence, the source "feels" the load through the coupling, and the current flow I_P is automatically ad-



FIG. 24. An audio frequency example of inductive coupling.

justed so that the required power is delivered, provided the coupler is properly designed. You will go into this in more detail in later lessons.

Practical A.F. Application. A very common and practical low-frequency application of inductive coupling is that shown in Fig. 24, where a microphone serves as a source of a.f. signals, an audio amplifier serves as a load, and an iron-core coupling transformer serves to transfer to the source the effects of the load. This coupling transformer can be designed to deliver maximum voltage or maximum power to the load.

Practical R.F. Application. A radio frequency application of inductive coupling is the little "pill-box" you see connected between the two halves of some short-wave antennas in an arrangement similar to that shown in Fig. 25A. A close-up view of this box



Fig. 25. An r.f. application of inductive coupling.

appears in Fig. 25B. Generally, all you will find inside it are two coils forming an air-core transformer, which is connected between the two halves of the antenna and the two-wire leadin (transmission line) in the manner shown in Fig. 25C.

▶ We might get a little in advance of our subject here to explain why this transformer is used. You know that an antenna has an impedance at its connecting terminals; likewise, the input of a radio receiver has impedance. As later lessons will show, maximum power can be transferred from one circuit to another only when their impedances are equal-or, as radio men say, matched. The simplest way to match impedances is to couple the circuits with a transformer which has a certain calculated number of turns in its primary and secondary windings. Thus, the transformer in Fig. 25B is used as an impedancematching transformer so that maximum signal power will be transferred from the antenna to the set.

Types of Inductive Coupling. The action of a coupling transformer, as

you already know, depends upon the fact that current flowing through the primary winding produces a varying magnetic flux. When all of the magnetic flux produced by the primary winding links with (passes through) the entire secondary winding, we have what is known as unity coupling. While unity coupling is just about impossible to secure in a practical transformer, a very close approach to it can be made; this is called tight coupling. On the other hand, when only a part of the magnetic flux links with the secondary winding, the coupling is said to be loose or weak, and the load will have less effect upon the source. If none of the magnetic flux links with the secondary winding, there is zero coupling. Let us consider how tight or weak coupling can be secured in actual coupling transformers.

Tight Coupling. When the two wires which form the primary and secondary windings of a coupling transformer are wound side by side on the coil form, as indicated in Fig. 26A, we have tight coupling; now practically all of the flux produced by the primary winding must link with or



FIG. 26. Unity coupling can be secured as shown here.

pass through the entire secondary winding. In ultra-high-frequency apparatus, the same result often is secured by using one or more turns of copper tubing as the primary coil, and. threading an insulated wire through this tubing to serve as the secondary coil. This is illustrated in Fig. 26B. When an iron core is used, as in Fig. 26C, tight coupling is secured without the necessity for having the turns of the two windings close together. In fact, the primary and secondary may be two separate coils placed one inside or alongside the other, as long as nearly all of the magnetic flux passes



FIG. 27. Loose coupling.

through both coils. Iron cores are used in coupling transformers designed for audio amplifier circuits.

Loose Coupling. Where loose coupling is desired, as in some r.f. transformers, the two coils generally are arranged in one of the two ways shown in Fig. 27, so that there is considerable space between the coils. Only a small fraction of the magnetic flux thus links with the secondary coil.

► When two coils are loosely coupled to each other, we really have a condition where a small part of the primary coil is *tightly coupled* with a small part of the secondary coil, and the remaining parts of the primary and secondary coils are not coupled to anything. These uncoupled parts of the primary and secondary coils can be considered as separate, extra coils in their respective circuits. Thus we can, if we wish, think of a loosely coupled circuit as being a tightly coupled circuit with an extra coil in its primary circuit and an extra coil in its secondary circuit. It is often very convenient to analyze a loosely coupled circuit this way. These extra inductances are known as the primary leakage inductance, since they exist because some magnetic flux leaks off without linking the other coil.

As you learn more about radio apparatus, you will see how important is this left-over or leakage inductance in the actual operation of radio circuits which employ loose coupling. Also, you will discover that moving the original coils closer together increases the coupling, reduces the values of these leakage inductances, and affects the operation of the circuit in many other ways.

Looking Forward. Now that we have a general understanding of resistors, coils, condensers, vacuum tubes, and tuned circuits, we can proceed to connect all these into actual vacuum tube circuits and study their operation in connection with tubes. This we will do in our next lesson, in which you will be introduced to a number of typical circuits actually being used in radio receivers and other radio apparatus. You will probably be surprised, after you have completed your study of this next lesson, to see how much you have learned from the first ten lessons in vour Course.

THE N. R. I. COURSE PREPARES YOU TO BECOME A
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Lesson Questions

Be sure to number your Answer Sheet 9FR-3.

Place your Student Number on every Answer Sheet.

Send in your set of answers for this lesson immediately after you finish them, as instructed in the Study Schedule. This will give you the greatest possible benefit from our speedy personal grading service.

- 1. Does reducing the resistance of a series resonant circuit: 1, increase; or 2, decrease the selectivity of the circuit?
- 2. When a series resonant circuit is tuned to resonance, is the circuit current flow: 1, at its highest value; or 2, at its lowest value?
- 3. Suppose the capacity of C in a series resonant circuit is increased. Will the circuit now tune to: 1, the same frequency; 2, a higher frequency; or 3, a lower frequency?
- 4. At resonance, does a parallel resonant circuit act like: 1, a coil; 2, a condenser; 3, a high resistance; or 4, a low resistance?
- 5. When not tuned to resonance, does a parallel resonant circuit act like the part: 1, having the *higher* reactance; or 2, the part having the *lower* reactance?
- 6. To obtain the highest possible selectivity, would you use a coil having: 1, a highQ; 2, a low Q; or 3, a medium Q?
- 7. What will happen to the inductance of an air-core coil when a powderediron core is placed inside the coil?
- 8. When a voltage is *induced* in the coil of a resonant circuit, is the circuit: 1, *series* resonant; or 2, *parallel* resonant?
- 9. Name the four methods of coupling an antenna to a resonant circuit.
 10. What condition is necessary for unity coupling?

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World Radio History

DETERMINATION

Did you ever watch a steamroller — an oldfashioned one with clanging gears, escaping steam and great clouds of black smoke belching from the stack? Wasn't it a thrill to see that great machine level out obstacles in its path, leaving a perfectly smooth roadway? The word "determination" always brings to my mind that picture of a steamroller —a machine which goes places once the "full steam ahead" lever is thrown.

So, too, are you determined to go places, to achieve success and happiness. One by one you are completing your lessons, studying hard and making sure you understand everything; step by step you are approaching that greatest of all goals—SUCCESS.

Of course, the way is long and not always easy. But whenever the going gets a little tougher than usual or you feel a bit discouraged, just bring out that old determination, and back it up with every single ounce of ambition you have.

Be a human steamroller, always confident of your own power, always moving ahead, always succeeding. Keep that determination of yours alive every single minute. I'm with you, ready to help you over the bumps at any time — I'm determined to make you win.

J. E. SMITH



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David W. Camif

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STUDY SCHEDULE NO. 10

For each study step, read the assigned pages first at your usual speed. Reread slowly one or more times. Finish with one quick reading to fix the important facts firmly in your mind, then answer the Lesson Questions for that step. Study each other step in this same way.

Now that you have learned the characteristics of radio parts, you are ready to combine them with tubes into radio stages. This section gives an introduction to the names and purposes of radio stages as they are found throughout a radio system. Use this "map" as you go along, to understand just how each kind of stage fits into the over-all picture. Answer Lesson Question 1.

Radio stages work because it is possible to arrange parts and voltages so that tubes will do certain things. The more you can learn about tubes, the easier you can understand stages. Answer Lesson Questions 2 and 3.

] 3. Basic Amplifying StagePages 13-18

All tube amplifying stages can be reduced to a simple, basic stage—au input device or signal source, a load to which the signal is delivered, and a power supply. Here are the details of how this basic circuit works; how changes in voltage or load value affect the operation; how signals are kept in their correct paths by by-pass condensers; how a tube can produce its own bias. This is a VERY IMPORTANT section. Read it over and over, until you feel you know everything in it. Then review it from time to time. Answer Lesson Questions 4, 5 and 6.

Now that you understand the basic stage, you are introduced to some of its variations. The major difference is in the type of coupling—that is, in the different ways of getting signals into and out of the stage.

] 5. Tubes as Signal Generators and as Signal Mixers . . Pages 23-28

Here is a brief introduction to radio stages which do other things in addition to amplifying signals. You are not expected to fully understand oscillators, modulated stages and frequency converters from this discussion, but you should learn what these stages do to the radio signal. Answer Lesson Question 7.

] 6. The Diode in Radio Systems Pages 28-33

Power supplies and demodulator stages commonly use the two-element (diode) tube. This introduction to these stages shows how it is used. Answer Lesson Questions 8, 9 and 10.

7. A Complete Superheterodyne Receiver Circuit ... Pages 33-36

A schematic diagram is used to show how radio signals progress through a superheterodyne receiver. This review of the lesson shows the manner in which all kinds of stages are brought together in a receiver—how each stage performs a necessary operation so that the reproducer can deliver the intelligence we want.

- 78. Mail your Answers for this Lesson to N.R.I. for Grading.
-] 9. Start Studying the Next Lesson.

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World Radio History

HOW TUBES WORK IN TYPICAL RADIO STAGES

How Radio Systems are Divided into Stages

THIS lesson is a milestone in your Course in Radio Fundamentals. Up to this time you have been learning the characteristics of *radio parts*. You will now begin to learn how these parts are combined into groups in which tubes are used—groups that are called *radio stages*.

In this lesson, we are going to do two things: 1, give you a preview of the radio stages you will study in detail in the next few lessons; and 2, give you a basic understanding of the requirements underlying ALL stages. Once you understand stages, it will be easy for you to combine them into sections and thus form a complete receiver or transmitter.

► To help you see the relationship between stages, we will first describe a complete radio system. Pay particular attention to the *names* and *functions* of the stages—and refer to this "radio map" from time to time so that you can keep these stages properly in mind. (Later in this lesson, we will give a description of a complete superheterodyne receiver in somewhat more detail.)

Naturally, we will not try to give every detail of the operation of radio stages in this one lesson. A preview of this kind is intended to give you enough basic facts to help you understand the full details to be found in later lessons.

RADIO SYSTEMS

All radio systems have one objectto convey intelligence from one point to another. The intelligence can be anything—a sound or sight (television) program; commercial (voice or code) messages going from point to point; a guide-path for aircraft or ships; or messages intended for police or fire departments.

The original intelligence cannot be carried to distant points without some form of conversion or without the use of a carrier. Even the loudest sounds cannot be heard more than a few



Courteey Galein Mfg. Corp.

This Motorola is typical of the console type radio receivers you are likely to meet if you become a serviceman. What is inside the cabinet? A superheterodyne receiver using the basic stages you are studying in this lesson. miles and close-up details of a scene are visible for only a few feet. We can record sounds or take photographs, but these are not instantaneous methods of transmission. It is also true that by converting sounds or scenes into electrical impulses it is possible to carry them over wire lines. However, at best, wire circuits have limits, so radio steps in and provides a means of sending intelligence instantly to any desired receiving point.

1

Radio transmission is made practical by the fact that high-frequency signals will travel through space, and it is possible to make these signals "carry" the intelligence. At the receiving point, a *copy* of the original intelligence signal can be obtained from the "carrier" signal and used to operate a reproducer system. Fig. 1 shows just such a system in block-diagram form. Let's review the basic actions in each stage.

THE TRANSMITTER

The Carrier Signal. At the transmitter, the "carrier" signal is generated by a vacuum tube stage known as an oscillator (indicated at A in Fig. 1). The carrier is called an r.f. (radio frequency) signal since it is in the r.f. range. At each station, the frequency of this signal is set at the value assigned to that particular station. An American broadcast station using the amplitude modulation (a.m.) system will be assigned a frequency within the 550 kc.-to-1600 kc. broadcast band, while an f.m. (frequency modulation) broadcast station will be assigned a carrier frequency somewhere within the 88- to 106-mc. range.

Amplified copies of this r.f. signal are obtained from a series of vacuum tube amplifying stages (in section Bof Fig. 1) which deliver the amount of power needed to make the proper com-

bination with the intelligence signal.

The Intelligence Signal. The intelligence (sound waves or visible scenes) is converted into a corresponding electrical signal by the pickup device at D. Amplified copies of this electrical signal are obtained from the amplifiers at E.

Modulation. At C, the carrier and the intelligence signals are combined. This process of combining is called modulation, because the intelligence signal modulates or varies the carrier signal. At present, two systems of modulation are in use: amplitude modulation (a.m.) and frequency modulation (f.m.). In amplitude modulation, the amplitude of the carrier signal is varied in proportion to the amplitude of the intelligence signal; in frequency modulation, the frequency of the carrier signal is varied in proportion to the amplitude of the intelligence signal. In either system, the rate at which variations in the carrier signal occur corresponds to the frequency of the intelligence signal.

The modulated carrier signal may be, and frequently is, amplified before being applied to the transmitting antenna or radiator G. In later lessons, we'll go into the subject of antennas and find why they radiate—here, all we'll say is that the signal is radiated into space in the form of a varying electromagnetic field which is known as a radio wave.

THE RECEIVER

When the radio wave encounters receiving antenna *H*, it induces a voltage in the antenna. This voltage causes a varying current flow through the lead-in; this, in turn, induces a voltage in the r.f. preselector *I*. All radio waves striking the antenna—whether they come from the desired station or from other stations—induce voltages in the r.f. preselector, so we use "tuned" or resonant circuits (coil-condenser combinations) in the preselector section to start the process of selecting the desired signal from all the others received. (Here you see the reason for assigning different frequencies to transmitters; only by frequency separation can we select the signals of the desired stations from among all others broadcasting at the same time.)

The preselector usually must tune over a wide range of frequencies, since, of course, we generally like to tune to different stations. Now, it is difficult to obtain the same amount of amplification and selectivity at different frequencies over wide ranges. It is far easier to obtain high gain and good selectivity from an amplifier operating at a *fixed* frequency, especially if this fixed radio frequency is in the lower r.f. range. These facts have led to the creation of the superheterodyne receiver, in which incoming signals, regardless of frequency, are converted into a fixed lower frequency for amplification.

The Superheterodyne. Briefly the fundamental principle of the superheterodyne circuit is: <u>The incoming</u> signal is combined with a "local" r.f.signal so that a lower frequency r.f. signal is produced.

Now, let's see what stages are used in a superheterodyne receiver to perform the action we have just described. In Fig. 1, preselector I is used to give an initial separation of the signals present in the receiving antenna H. The preselector is not required to give much amplification nor much selectivity; these are obtained after the frequency mixing has occurred. The mixer-first detector stage J is fed the modulated r.f. signal from the preselector and is also fed another (unmodulated) r.f. signal from the local r.f. oscillator stage K. (This is called a "local"



FIG. 1. A complete radio system. After the sound or picture signal is combined with the r.f. carrier at the transmitter, the combined signal is radiated The superheterodyne receiver selects the desired signal, amplifies and demodulates it, then amplifies the intelligence signal further so as to operate the reproducer. This block diagram applies to both a.m. and f.m. sound systems as well as to television. The major differences are in the r.f. and intelligence frequencies handled, and in through space by the antenna system G. At the receiving location, the radio wave is picked up by receiving antenna H. the types of modulation and demodulation stages. Hence, this diagram can be applied to ALL radio systems,

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oscillator to distinguish it from the oscillator in the transmitter.)

The result of the frequency combination (called "frequency conversion") is a new-carrier which has a frequency known as the *intermediate fre*quency (abbreviated i.f.) and which is modulated by the intelligence signal. The i.f. is usually equal to the difference between the carrier frequency and the local r.f.*

► The preselector and the local oscillator are tuned simultaneously by a single control which rotates their tuning condensers or varies their coil inductances in step. Thus, as we tune to a particular station, the local oscillator is automatically adjusted to the frequency necessary to produce the desired combination.

The tuned circuits at the input of the i.f. amplifier (L in Fig. 1) tend to select the desired i.f. carrier (carrying the intelligence signal) from among all others offered by the mixer-first detector. Each stage in the i.f. amplifier makes an amplified copy of the desired signal, and the resonant circuits in each stage serve to further select the desired signal from among others.

When the modulated i.f. carrier signal has been amplified to the limits of the i.f. amplifier, it is fed into a *demodulator* stage. (This stage is another *detector*, and is called the *second detector* to distinguish it from the *mixer-first* detector.)

Demodulation. The demodulator M serves a very important purpose; it is here that the intelligence signal is

separated from the carrier. The carrier has been necessary to convey the intelligence signal through space. The intelligence signal must remain with the carrier-until the desired-carrier canbe selected from among all others. Then, when the carrier has served its purpose, it is cast aside and a reasonably accurate duplicate of the original electrical intelligence signal is obtained.

The intelligence signal is similar to that existing at D in the transmitter, and usually must be amplified further before it can operate the reproducer. The intelligence signal amplifier N is called an *audio* amplifier if the signal corresponds to sound, or is called a *video amplifier* if the signal corresponds to a television picture scene.

Reproduction. After the intelligence signal has been sufficiently amplified, it must pass through a reproducer O which will convert the electrical signal into the original form of the intelligence. For audio signals (voice or music), a loudspeaker is This device produces sound used. waves in accordance with the variations in electrical current. For television, an image reconstructor tube "naints" the television scene on a screen at one end of the tube. Code signals may operate a sounder or an automatic typewriter. Facsimile signals may operate a device which chemically reproduces a picture on special paper. And so it goes-whatever the original signal, a suitable device restores the original sound, scene, or impulse into a recognizable or usable form.

► This completes the journey of our intelligence signal. It has travelled through many radio stages at both the transmitter and the receiver. Most of the stages made amplified copies, which in turn were amplified by suc-

^{*}Here is an example: If the local oscillator of a receiver generates a 1460-kc. signal when the r.f. section of the receiver is tuned to a 1000-kc. station, the i.f. will be 460 kc. (1460 — 1000 = 460 kc.). This is called the i.f. value because it is normally *intermediate* in value between the r.f. carrier frequency and the frequencies of the intelligence signal.

ceeding stages. Some stages did nothing but amplify, others performed specialized duties, while still others did both. This has been just a bare outline of the process of radio transmission and reception—we have not tried to show the *number* of stages in each section nor have we shown all that may be done in some of the more elaborate systems. Instead, we have given you a brief over-all picture to make it easier for you to see the relationships between the *stages* you are about to study.

Importance of Tubes. One important fact stands out about these stages: whatever their function, vacuum tubes are used in all of them, except possibly the intelligence pickup D and the reproducer O. (In a television system, special tubes are used at D and O too.)

Because of this widespread use of tubes, you must have a thorough understanding of fundamental tube operation before you study stages. We will now expand some of the information presented in your earlier lesson on tubes, thereby giving you a more complete picture of tube action under different operating conditions. Then we will be ready to study the operation of each basic radio stage.

More Facts About Tubes

There are many kinds of tubessome designed for specific purposes and some for general purposes. Basically, they all operate over rather similar characteristic curves; the particular actions obtained depend upon the operating point chosen and upon the devices used as "loads." Although tubes with two, three, and four grids are better than triode (single grid) tubes for certain purposes, we will consider only diode and triode tubes in this lesson. This will let you concentrate on the important features of the stages described without becoming confused by extra grids. (You will take up multi-grid tubes as you study each stage in detail in later lessons.)

HOW THE GRID CONTROLS THE PLATE CURRENT

From your earlier studies you know that, in a triode tube with a fixed plate voltage, the grid voltage determines the amount of plate current which will flow at any particular instant. When no signal voltage is applied to the grid, the bias voltage (a d.c. voltage placed between the grid and the cathode) determines the amount of plate current flow; the plate current which flows under these conditions is often called the "operating" current. (This value is the initial or starting plate current which is then varied by the signal.)

In most receiver circuits the grid is not allowed to become positive with respect to the cathode, because this would cause an undesirable flow of grid current. (In transmitter r.f. power amplifier circuits and in oscillator circuits, the grid is allowed to go positive, as we shall see later.) Hence, we usually have a *negative* grid bias. The amount of bias depends on the point at which we desire to operate on the characteristic curve.

A typical characteristic curve is shown in Fig. 2. For a tube which has this characteristic, a negative bias of -3 volts places the operating point at


FIG. 2. The characteristic curve for a 6P5G tube. Although characteristic curves are similar in shape, the values apply to only one tube type and to one particular supply voltage value.

A (on the straight portion of the curve). A bias of -8 volts causes the operating point to move to B, while a bias of more than -9 volts will "cut off" the plate current. If a different plate voltage is used, then the curve will "move" so that other bias values will be required for these conditions. For this reason, curves must be labeled as to what plate voltage is being applied.

► Naturally, tubes differ in their plate currents and bias requirements, although the general *shape* of their characteristic curves may be very similar.



FIG. 3. Typical bias and plate voltage connections are shown here. However, the filament supply is NOT shown. Naturally it IS connected to the filament terminals and must be of the proper value to produce temperature saturation. Its omission on diagrams is purely to simplify the circuit so that you can concentrate on the actions being described.

Another tube may require a bias of -50 volts to operate at point A, and may then pass a current of 30 ma. instead of the 6-ma. value shown in Fig. \mathscr{D} . Thus, the same curve may fit both tubes, except that the values of current and voltage differ. Therefore, characteristic curves must be labeled also for the particular tube to which they apply.

▶ Now, with our operating point determined by some suitable bias, let's see what happens when we apply a sine-wave signal to the circuit shown in Fig. 3.



FIG. 4. The E_F-1_p curve for a 6J5 tube, with a plate voltage of 250 volts. It is assumed that the proper filament voltage is applied to the tube.

Fig. 4 is an E_{g} - I_{p} (grid voltage-plate current) curve for the triode tube used in Fig. 3. With the plate and bias voltages shown in Fig. 3 (which are typical for this tube), the operating point is at S, and the operating plate current value is nearly 9 ma. We know that the applied signal makes the grid voltage alternately more and less negative than the grid bias, so when the signal shown at A (in Fig. 4) is applied, the grid voltage will be made up of the bias value x plus or minus the value of the signal voltage y at that instant. Since the bias voltage is greater than the largest value of the signal voltage, the grid remains negative at all times and there is no grid current flow. The grid is purely a *potential-operated device* its *voltage* at any instant determines the plate current value.

Let's see how a change in the grid voltage affects the plate current. Suppose the grid voltage changes from point 1 to point 2 on curve A (making the grid less negative by 3 volts). Tracing vertically from each of these points up to the E_g - I_p curve, then horizontally over to curve B, we find that this grid voltage change causes a plate current change from point 1 to point 2 on curve B. The plate current changes from its operating value of 9 ma. (point 1) to a value of about 15.5 ma. (point 2)—an increase of about 6.5 ma.

Similarly, changes in grid voltage from point 2 to point 3, from point 3to point 4, and from point 4 to point 5on curve A cause plate current changes as shown by similarly-numbered points on curve B. Each grid voltage swing in a less negative direction causes an *increase* in plate current, while each swing in a *more* negative direction causes a *decrease* in plate current.

Another important fact is that the plate current is a *pulsating d.c.;* that is, it is made up of a d.c. component v (equal to the no-signal operating current) and a sine-wave a.c. component w which causes the total plate current to vary above and below its no-signal operating value. (The plate current cannot reverse its *direction* of flow; hence, the a.c. variation can exist only as a *component* or part of the pulsating plate current. When we speak of an a.c. plate current, we mean *only* this varying component, not the total plate current.)

► The signal actually applied to an amplifier rarely has a sine-wave shape; it usually is more complex. However,

a complex signal is made up of many different sine-wave voltages in combination, so the tube response to a sinewave signal corresponds to its response to that component of a complex signal. This makes it possible to use sine-wave signals when studying radio circuits. However, as Fig. 5 shows, if we know the exact wave shape of a complex signal, we can prove that the plate current has the same form as the grid voltage, provided that the E_g - I_p curve is linear (straight) over the operating region, as is the section M-N of Figs. 4 and 5.



FIG. 5. The signal currents handled by most radio and television tubes have irregular wave forms much like that shown here for e_g . It can be proven by mathematics, however, that any wave, no matter how complicated, can be broken up into a number of simple sine waves. We can simplify our work by considering only sine-wave signals, for we know that radio circuits will behave the same for a particular sine wave as they do for

this component of a complex wave.

CLASSES OF AMPLIFIERS

Not all amplifying stages operate on the straight portion of their characteristic curves; sometimes an exact "carbon copy" of the grid voltage is not wanted. This means that amplifiers must be classified according to the part of the characteristic over which they operate.

► When an exact duplicate of the en-



FIG. 6. In class B amplifiers, only half of the incoming signal is effective in producing plate current pulses.

tire input signal is wanted, the operating point is chosen near the center of the straight portion between the zero grid-voltage value and the lower bend or curved portion of the characteristic curve, and operation is confined to this straight portion. This is called class A amplification. Class A amplification is relatively distortionless (the output is practically proportional to the input at all times) provided: 1, that the signal voltage is not so large that some curved portion of the characteristic is encountered; and 2, that the grid never goes positive.

For some purposes, we do not want the plate current variations to be exactly like the *entire* signal applied to the grid circuit. We may, for example, apply a bias sufficient to cut off plate current, as shown in Fig. 6. Now when we apply a signal, we will get halfwave pulses in the plate circuit—in other words, very little (if any) plate current will flow until the positive signal alternations overcome the bias. This is called *class B operation* and is very necessary for certain purposes.

► Going a step further, we may even apply a bias many times that required for plate current cut-off, as shown in

Fig. 7. Only the positive tips of the grid signal can cause plate current to flow. This is called class C operation. If we allow the grid signal to swing over into the positive region (Figs. 6 and 7), class B and class C amplifiers become far more efficient than class A (that is, they produce more output for the same plate supply power). However, the resulting distortion generally limits such operation to special stages where the distortion can be overcome or does not matter. We shall explain these uses in later lessons.

FAMILIES OF CURVES

In practical radio work, tables of tube characteristics furnish practically all the needed information, so curves are rarely used. However, characteristic curves are extremely valuable to you now as a *student* in helping you to understand just how tubes work—so let's take a few minutes to see how these curves are plotted and what they reveal.

Plotting Curves. To make curves, the tube manufacturer uses a circuit similar to that shown in Fig. 8. This circuit is arranged so that any desired C voltage can be applied, the plate voltage is variable, and the plate current can be measured. Normal fila-



FIG. 7. In class C operation, even less than half of the incoming signal is used to produce plate current pulses.

ment voltage is applied.

To make E_{g} - I_{p} curves, the first step is to choose some value of plate voltage, by adjusting tap P_{2} on the B battery. Then the grid voltage is varied in steps from zero to a maximum negative value by moving the slider from terminal 1 toward terminal 2 on the potentiometer P_{1} in the grid circuit. The plate current is measured and plotted for each particular grid voltage value. The resulting curve is the E_{g} - I_{p} characteristic curve of the tube for that particular plate voltage.

This process is repeated for other values of plate voltage until a series or family of $E_{\rm g}$ - $I_{\rm p}$ characteristic curves is secured. They are similar to those shown in Fig. 9 (which are for a 6C5 tube).

► Although a family of curves can be made to tell almost everything about the behavior of a tube, radio men compress some of the most important information in characteristic curves into quickly usable forms. Ratings like horsepower, maximum speed, and gasoline consumption tell much about the performance of an automobile; tube performance is likewise expressed by the following ratings:

- 1. Amplification factor.
- 2. A.C. plate resistance.
- 3. Mutual conductance.

You've already met the first two of these in an earlier lesson. We'll review them briefly, then discuss the third rating (which is really a com-







FIG. 9. A family of characteristic curves.

bination of the other two).

AMPLIFICATION FACTOR

This factor is important because it tells us how good a tube is as a voltage amplifier.

Suppose we use Fig. 10, showing two curves taken from Fig. 9, to illustrate this. If the 6C5 tube is operating at point a on curve B (for a plate voltage of 200 volts), then a plate current increase of 4 ma. can be obtained by changing the grid voltage 2.5 volts (from -8.5 to -6 volts); this moves the operating point to c. Similarly, with the bias voltage fixed at --8.5 volts, a 4-ma. change can be obtained by increasing the plate voltage 50 volts, from a on curve B to b on curve A. (Curve A is for 250 volts, which is 50 volts higher than curve B.) We can thus get a certain change in plate current either by leaving the plate voltage fixed and changing the bias, or by leaving the bias fixed and changing the plate voltage. However, notice that the plate voltage change is far larger than the grid voltage change needed to give the same plate current change.

If we now divide 50 (plate voltage change) by 2.5 (grid voltage change) we get 20. This means that the grid is twenty times as effective as the plate

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(for this particular tube) in controlling the plate current. This value is called the *amplification factor*, and in plain language it means that, for this tube, a 1-volt grid change will affect the plate current just as much as will a 20-volt plate-voltage change. In other words, a 1-volt signal on the grid can act like a 20-volt variation in the plate circuit of this tube.

► The amplification factor does not tell directly how much the grid controls plate current, but it does express the relative effects of the grid and plate voltages on the plate current. The amplification factor can be defined technically as follows:

Amplification factor is equal to the plate voltage change which will produce a certain plate current change, divided by the grid voltage change which will produce the same plate current change.

Amplification factor, also called amplification ability, amplification number, or amplification constant. is generally designated by the Greek letter mu (μ) , which is pronounced "mew." Amplification factor depends mostly upon the construction of the tube --- particularly upon the position of the grid between the cathode and plate and upon the spacing between the meshes or turns of the grid wire. The closer the grid is to the cathode and the closer together are the grid wires, the larger will be the μ of the tube, because the grid then will have a greater control over the electron movements. \blacktriangleright The values of μ given in tube characteristics charts are only averages, as tubes of the same type may vary as much as 20% from the rated value. This deviation applies to other tube ratings as well, for radio tubes are delicate devices, made so compactly that errors of a few thousandths of an inch in the position of an electrode will greatly affect the tube ratings. These

differences in tube characteristics are permitted because close similarity is generally not necessary.

A.C. PLATE RESISTANCE

The opposition which the platecathode path of a tube offers to the flow of a.c. is expressed in ohms, just as is any other resistance, and is commonly abbreviated as $r_{\rm p}$.

The value of r_p for a tube is found by applying to the plate-cathode an



FIG. 10. Two of the curves taken from the family in Fig. 9.

a.c. voltage and measuring the resulting change in plate current, or it can be found from the characteristic curves. For example, referring to Fig. 10, a plate voltage change of 50 volts (a to b) produces a plate current change of 4 ma. when the tube is biased at -8.5 volts. As we can consider the change to be equivalent to an a.c. variation, we need only use Ohm's Law to find the a.c. resistance. In this case, then, r_p is equal to 50 \div .004, or 12,500 ohms. Notice—this value applies only

D.C. Plate Resistance. A radio tube is peculiar in that it offers a different plate-to-cathode resistance to a steady voltage than it does to a varying voltage. When only d.c. voltage is applied to the grid and the plate, a definite amount of d.c. plate current will flow. By dividing the d.c. plate voltage by the d.c. plate current (Ohm's Law), we get the d.c. plate resistance of the tube for that particular grid bias value. Thus, at point a in Fig. 10, the d.c. plate voltage is 200 volts and the d.c. current is 3 ma., so the d.c. resistance is $200 \div .003$, which is about 66,000 ohms. (This is for the chosen value of grid bias only-since changing the grid bias changes the plate current and therefore changes the d.c. plate resistance, the bias must always be specified when a value of d.c. plate resistance is given.) This is quite different from the a.c. resistance.

► Strictly speaking, the designations "a.c." or "d.c." should always be given when the name "plate resistance" is used. However, for amplifiers, the a.c. behavior of a tube is the most important, for it tells us what the tube will do when signals are applied; d.c. resistance is seldom of interest. Whenever you see the plate resistance of a tube mentioned, you can assume the a.c. value is meant, unless it is definitely marked "d.c."

MUTUAL CONDUCTANCE

The third important tube rating, called mutual conductance or transconductance, is equal to the amplification factor divided by the a.c. plate resistance. This rating tells directly how much the grid voltage controls the plate current; it is defined as follows: Mutual conductance of a tube is equal to the a.c. plate current divided by the a.c. grid voltage, when there is no load in the plate circuit.

Mutual conductance, abbreviated as $g_{\rm m}$ (or $s_{\rm m}$), is expressed in *mhos* (pronounced "mose"). If we divide the a.c. plate current (the varying portion of the plate current) in amperes by the a.c. grid voltage in volts, we obtain mhos. However, the mutual conductance of tubes is but a small fraction of a mho, so, to avoid an awkward decimal number, the *micromho*, equal to one-millionth of a mho, is used in radio work. (Incidentally, "mho" is "ohm" spelled backwards.)

Mutual conductance (sometimes known as transconductance) shows how much a.c. plate current will be produced by an a.c. signal in the grid circuit. If the grid voltage in volts and the mutual conductance in mhos are known for a tube circuit, such as an amplifier tube circuit whose load has negligible resistance, the a.c. plate current in amperes is obtained simply by multiplying the two values together.

When a tube ages, its electron emission is lowered and its g_m is reduced. This is why g_m is often measured when testing radio tubes, because it indicates the worth of the tube.

► Values of mutual conductance are given on tube charts. They can also be obtained from the E_g - I_p curves. In Fig. 10, a grid voltage change of 2.5 volts (from a to c) produces a plate current change of .004 ampere. Since a change corresponds to an a.c. value, the mutual conductance of the 6C5 tube (for operating point d in the middle of the operating region of curve B) will be .004 ÷ 2.5 or .0016 mhos. This is the same as 1600 micromhos.

An example of a tube chart is given in Fig. 11. Note that r_p , g_m , and μ vary greatly among the tubes listed.

EQUIVALENT TUBE CIRCUITS

The tube ratings we have discussed so far are those of the tube alone, with no load in the plate circuit. These ratings are useful, but they do not tell directly how much amplification can be obtained in a practical stage, where a load must be used. When we wish to find the *actual stage gain*, it is far simpler to consider an equivalent circuit (such as was shown in an earlier lesson). This circuit deals only with the a.c. voltages and the a.c. currents in the plate circuit of the tube.

Equivalent A.C. Plate Voltage. First, let us explain what is meant by equivalent a.c. plate voltage. You know a.c. grid voltage in its effects.

We use the tube amplification factor to find this equivalent voltage. Since this factor shows how many timesmore effective is a change in grid voltage than is a change in plate voltage, the plate voltage change is equal to the grid voltage change multiplied by the μ of the tube. Hence, we say that the **a.c.** grid voltage e_{z} multiplied by the μ of the tube is the equivalent a.c. plate voltage μe_{z} . This equivalent a.c. plate voltage is used in calculations just as if it were an actual voltage; in fact, most engineers speak of it as the a.c. plate voltage.

The Circuit Without a Load. Now, let us draw the equivalent circuit of an

TYPE	CLASS	BASE	FILAMENT RATING		HEF	PLATE	NEGATIVE	SCREEN	PLATE	SCREEN	PLATE RESISTANCE	9m MICROMHOS MUTUAL	H AMPLI- FICATION	OHMS LOAD FOR	UNDIS- TORTED POWER
			VOLTS	AMPS	036	VOLTS	VOLTS	VOLTS	MA,	MA.	OHMS	ANCE	FACTOR	POWER	WATTS
SK5GT/G	TRIODE	5-U	6.3	0.30	AMPLIFIER	100 250	1.5 3.0		0.35		78,000 50,000	900 I,400	70 70		
SK6GT/G	PENTODE	7-5	6.3	0.40	POWER AMP.	100 250 315	7.0 18.0 21.0	100 250 250	9.Ó 32 0 25 5	1.6 5.5 4.0	104,000 68,000 75,000	1,500 2,300 2,100		12,000 7,600 9,000	350 3,400 4,500
6K7GT/G	PENTODE	7-R	6.3	0.30	AMPLIFIER	90 250	30 3.0	90.0 100	5.4 7.0	1.3	300,000 + 800,000	1,275	400 \$160		
6L5G	TRIODE	6-Q	6.3	0.15	AMPLIFIER	100 250	3.0 9.0		40		10,000 9,000	1,500	15		
6L6	BEAM AMP.	7-AC	6.3	0.90	POWER AMP.	250 350	14.0 18 0	250 250	72 0 54.0	50 2.5	22,500 33,000	6,000 5,200		2,500 4,200	6,500 10,800
607	OUODIODE TRIODE	7-V	6.3	0.30	DET. AMP.	100 250	1.5 3.0		+ 0.35 I.F		88,000 58,000	800 1,200	70 70		

FIG. 11. A section of a typical tube chart. Reading from left to right we find that the columns give: 1, the type number on the tube, assigned according to the RMA code; 2, the class of tube, whether a triode, pentode, etc.; 3, the base connections are given in another section of the chart (not shown here) but the key letter and number in column 3 refer to similarly-keyed base views; 4 and 5, the filament voltage and current; 6, the use to which the tube can be put; 7 through 11, the voltages and currents of the various plate, screen grid and control grid electrodes; 12, 13 and 14, the plate grid resistance, mutual conductance and amplification factor; 15, the proper load value; 16, the power output. This chart only shows the characteristics of a few of the many hundreds of different tubes used in modern radio receivers. Notice how even these tubes vary widely in characteristics.

that an a.c. grid voltage, or signal, acting in one circuit of a tube (the grid circuit) produces an a.c. plate current in another circuit (the plate circuit). Thus, we are dealing with two different circuits connected only by the action of a tube. It is always simpler to study a single circuit at a time. For this reason, we transfer the *effects* of the a.c. grid voltage to the plate circuit; we forget the grid circuit and assume that there is only a plate circuit containing an a.c. plate voltage *equivalent* to the amplifier having no plate load. Considering only the a.c. actions, we find that the only opposition to the flow of current in the plate circuit is the a.c. plate resistance r_p , and that the only a.c. voltage acting in the circuit is the equivalent a.c. plate voltage μe_g .

Therefore, we can consider the a.c. plate circuit to be as shown in Fig. 12A — an a.c. source and a resistance r_p . The a.c. plate current flows through r_p and all the equivalent a.c. voltage is dropped in this internal tube resistance.



FIG. 12. Equivalents of plate circuits with and without a load.

Equivalent Tube Circuit With Load. The circuit without a load is useless, as all the signal energy is lost in the tube plate resistance. Hence, in a practical amplifier, we have to use a load in the plate circuit so that some of the signal energy can be developed outside the tube where it can be put to use. This load $R_{\rm L}$ is in series with the resistance $r_{\rm p}$, as shown in the equivalent circuit of Fig. 12B. The insertion of this load will not change the operation of the tube, provided the same d.c. operating voltages are still applied to the tube. (If the load is a resistor and we want to operate the tube at a definite plate-to-cathode d.c. voltage, then the supply must furnish this plate voltage

plus the drop which occurs because of the d.c. plate current flow through the load resistor. The values of μ and r_p will then be those shown in the tube characteristics chart.)

With a load in the plate circuit, the equivalent a.c. plate voltage is divided between r_p and R_L .^{*} That portion of the a.c. plate voltage which is developed across R_L is the useful portion of the signal voltage and can be applied to other circuits.

*For example, let's suppose e_r is 2 volts; μ is 10; r_p is 10,000 ohms; and R_L is 50,000 ohms. The voltage division will be in the ratio of R_L to the total $(R_L + r_p)$ resistance, so the amplified signal voltage developed across R_L will be:

$$= \mu e_{\epsilon} \times \frac{R_{L}}{R_{L} + r_{p}}$$

= (10×2) × $\frac{50,000}{50,000 + 10,000}$
= 20 × $\frac{50,000}{60,000}$
= 20 × $\frac{5}{6}$ = 16.66 volts

Thus, in this case, the original 2-volt signal has produced a voltage of about 17 volts across the load resistor. With a μ of 10, the value of μe_s is 20, but we can never get all this as there is always some loss across $r_{a.}$

The Basic Amplifying Stage

Any amplifier, whatever its purpose, can be reduced to the simple circuit shown in Fig. 13. As shown here, the basic requirements for a tube amplifier are: 1, a signal source; 2, the proper operating voltages; 3, a load to which the signal is delivered.

Signal Source. The signal source which feeds into the grid-cathode terminals of a tube may be any device capable of supplying the desired signal voltage. The basic operation of a tube stage will be the same regardless of the source—whether the signal source is a photoelectric cell, a microphone, or a television camera tube, or whether the signal is obtained from a preceding stage.

Amplifier Load. The load is the device to which the tube *delivers* a copy of the voltage applied to the grid. As there must be a path for the d.c. supply voltage to the tube, the load (or a coupling part) will be a resistor, a coil, or a transformer.

Supply Voltages. We have already

discussed the importance of supply voltages in this and other lessons. The plate voltage and grid voltage are chosen to cause operation at the particular point on the characteristic curve desired for that particular amplifier. Under those conditions, the applied signal will cause the proper variation in the plate current to give the desired output voltage across the load.

VOLTAGE AND POWER AMPLIFICATION

Amplifiers can have one of two purposes: 1, if an amplifier makes the signal voltage across its load as large as possible (much larger than the grid signal voltage), it is called a voltage amplifier; or $\cdot 2$, if an amplifier makes the signal power in the load as large as possible, it is called a power amplifier. In this latter case, the a.c. load current multiplied by the a.c. load voltage should be as large as possible.

Voltage amplifiers are needed to build up the strength of weak signals,





while power amplifiers are needed to operate most reproducers and to build up the power at the transmitter to the desired level. The basic circuit shown in Fig. 13 will be the same in either case—we can get either maximum signal voltage or maximum signal power by the proper choice of a tube and by adjusting the value of the load.

To see just how the load can have this effect, let's go back to our equivalent circuit shown in Fig. 12B. To get the greatest voltage amplification, we



FIG. 14. These curves show the effect that various load values have on the plate current, load voltage, and power output. The higher the load value the greater the VOLTAGE gain. The load should equal the source impedance for greatest POWER output.

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want V_L (i_p times R_L) to be as large as possible, while for power amplification we want the power absorbed by the load (V_L times i_p) to be as large as possible. To illustrate the effects of the load resistor, let's assume that r_p is 10,000 ohms, that μ is 10, and the a.c. grid voltage is 1 volt.

A.C. Plate Current. From Ohm's Law, the a.c. plate voltage divided by the total a.c. resistance in the plate circuit gives the a.c. plate current i_p . (This a.c. plate current is only a part of the *total* plate current—it is just the variation caused by the signal. We are not interested in the d.c. current, so we ignore it.) By assuming different values of R_L and plotting the resulting plate current^{*} we would get the curve A of Fig. 14. As we expect, increasing R_L (that is, increasing the total resistance) causes i_p to be reduced.

A.C. Output Voltage. The load voltage is simply the product of i_p and R_L . By multiplying together these values for each size of load resistor, we will get curve *B* of Fig. 14. Although i_p is decreased by larger values of R_L , their product is increasingly larger that is, R_L increases faster than i_p decreases.[‡]

This voltage across R_{L} is the output voltage; the larger R_{L} is the larger the output voltage becomes. The true stage gain is equal to the output divided by the input, and since we chose an input

*The a.c. plate voltage is μe_{g} , and the total resistance is $R_{L} + r_{p}$, so

$$i_{p} = \frac{\mu e_{g}}{R_{L} + r_{p}}$$

‡We could arrive at the same result by remembering that the a.c. plate voltage $\mu e_{\rm g}$ divides between $R_{\rm L}$ and the plate resistance $r_{\rm p}$; as $R_{\rm L}$ is increased, a greater part of the a.c. plate voltage is developed across it. In fact, we can calculate the load voltage for any particular input voltage from the relation:

$$V_{\rm L} = \mu e_{\rm g} imes rac{R_{\rm L}}{R_{\rm L} + r_{\rm g}}$$

grid voltage of 1 volt, in this particular case the voltage V_L is equal to the true voltage amplification.

Power Output. The load power is the product of i_p and V_L —the a.c. plate current multiplied by the a.c. voltage across the load. By multiplying together the values obtained on curves A and B of Fig. 14, we get curve C, which represents the load power for various values of load resistance. There are a number of very important facts revealed by the three curves in Fig. 14. These are the points you should remember:

1. Effect of Load on A.C. Plate Current: The insertion of a load in the plate circuit of an amplifier tube decreases the a.c. plate current; the greater the ohmic value of the load, the less the current.^{‡‡} When the plate load resistance is exactly equal to the a.c. plate resistance, the a.c. plate current will be reduced to exactly half its no-load value (the value when R_L is not in the circuit).

2. Effect of Load on Amplification: Increasing the plate load resistance increases the stage gain, so the maximum gain is obtained when $R_{\rm L}$ is many times larger than r_p . The maximum possible amplification is μ , the amplification factor of a tube, but this limit can never be reached in actual practice, since there is always some loss in voltage across r_p . When R_L is equal to r_p , only 50% or one-half the total possible amplification is obtained, and when $R_{\rm L}$ is nine times the value of $r_{\rm p}$, about 90% of the maximum amplification is obtained. (Practical voltage amplifying stages using triode tubes usually have load values about 7 to 10 times the value of $r_{\rm p}$.)

^{‡‡}It is assumed that the supply voltage is adjusted so the d.c. plate-cathode voltage and the d.c. plate current do not change.

3. When Maximum Power is Delivered to the Load. The load in a triode tube amplifier gets the maximum power when the resistance of the load is equal to the a.c. plate resistance of the tube; the power then decreases gradually as $R_{\rm L}$ is further increased. Thus, maximum power output is obtained when the load resistance equals or matches the a.c. plate resistance. From this, you can see that the value of plate load resistance used in a stage is important. In your service work, you will be called upon many times to replace the part used as a load. Naturally, the replacement part should be approximately the same value as the original, although generally you can



FIG. 15. How the effects of the power supply resistance are eliminated by a by-pass condenser.

use a somewhat higher value if an exact replacement part cannot be obtained.

► Incidentally, maximum undistorted power output is not obtained with the same load value as is maximum power, as we shall see later. With triode tubes, distortion is less when the load value is about twice the a.c. plate resistance. Tube charts give load values for maximum undistorted power output.

BY-PASS CONDENSERS

Returning to the circuit of Fig. 13, you will notice that condensers C_1 and C_2 are connected across the batteries. Now, batteries have some internal resistance and, as they grow older, this resistance increases. Also, all power supplies (used in place of batteries) have considerable impedance. If the signal current has to flow through this impedance or resistance, several undesirable actions will occur.

The equivalent circuit containing this added resistance is shown in Fig. 15A. Obviously, the added R_s increases the total resistance, thus reducing i_p and reducing the output voltage (i_p times R_L) across R_L .

We can minimize this effect by connecting a condenser C_1 across the power supply, as shown in Fig. 15B. By choosing a capacity large enough to have a very small reactance to the signal current, we find that the condenser will act practically as a short circuit for signals around $R_{\rm s}$, so that little signal energy is lost in R_s . We can then ignore R_s and can consider the a.c. circuit to be $r_{p}-R_{L}-C_{1}$, with the opposition of C_1 so low that the circuit is, for all practical purposes, the same as was shown in Fig. 12B. (Of course, the condenser, if in good condition, does not pass appreciable d.c., so the d.c. operating voltages are still supplied to the tube through $R_{\rm L}$ in the usual manner.) Because C_1 offers another path for a.c. around the undesired path, it is called a by-pass condenser.

► There is another reason for keeping the signals out of the power supply. Usually the power supply furnishes d.c. voltage to more than one stage. If we let signals get into the power supply, they may flow into the other stages powered by the supply. Such coupling is undesirable—it may cause squealing and howling, or it may reduce the gain; you will learn about this in later lessons.

► Finally, we don't want stray a.c. voltages to come from the supply, along with the desired d.c. The same

by-pass condenser helps here too. If we represent the unwanted a.c. in the supply by generator e in Fig. 16, you can see that there are two paths for its current; 1, through $R_{s}-R_{L}-r_{p}$; and 2, through $R_{s}-C_{1}$. As C_{1} has a low reactance, it forms a filter with R_{s} , such that most of the undesired a.c. is dropped in R_{s} . Thus, the condenser again acts practically like an a.c. short circuit, reducing the a.c. voltage between terminals 1 and 2 to the point



FIG. 16. Stray a.c. voltages from the power supply do not affect the rest of the circuit if the by pass condenser C_1 is in good condition.

where very little undesired current will flow through $R_{\rm L}$ and $r_{\rm p}$.

These reasons for using a by-pass condenser across the power supply in the plate circuit also apply to the grid circuit, where a by-pass condenser generally is used across the d.c. bias supply. Condenser C_1 in Fig. 13 does not interfere with the d.c. bias supply, but does prevent a signal voltage drop across the supply; it also prevents any a.c. voltage in the bias supply from affecting the grid circuit.

► Thus, by-pass condensers across operating voltage supplies serve three important purposes:

- 1. They prevent an undesirable loss of signal.
- 2. They prevent undesirable coupling between stages.
- 3. They prevent undesired signals from the power supply from getting into the amplifying stage.

► Should a by-pass condenser become defective, the stage operation is certain

to be upset. If the condenser open-circuits (perhaps because a lead pulls away from a foil plate), then the condenser is in effect "not there." We will consequently get a loss of gain, undesirable coupling between stages, and perhaps hum or noise voltages from the power supply. On the other hand, if the condenser short-circuits, it will provide an unwanted path for d.c. This "short" would act like a direct wire connection between terminals 1 and 2of Fig. 16, so it would remove all d.c. plate voltage and cause a "dead" stage.

SELF-BIAS

The amplifier circuit in Fig. 13 shows a C battery as the grid bias supply. However, if the radio is equipped



FIG. 17. The tube can furnish its own C bias when this circuit is used. Plate current moving from B— to the cathode causes a voltage drop across R₁. As this voltage is between the grid and the cathode, it acts as the blas.

with a power pack so it can operate from a power line (as most sets are today), obviously we don't want to have to use batteries too. While it is possible to obtain bias voltages from the power pack, it is not always convenient to do so, particularly if the pack must supply a number of different stages.

World Radio History

The circuit shown in Fig. 17A makes it unnecessary to use any external scurce to furnish the proper d.c. bias voltage for the tube. In this circuit, the electron flow in the plate circuit is from the cathode of the tube to the plate, through $R_{\rm L}$, through the plate power supply (B+ to B-), and then through resistor R_1 back to the cathode. Thus, the d.c. plate current must flow through R_1 .

This current flow through R_1 produces a d.c. voltage drop across it. The polarity of this voltage drop will be as shown, because electrons enter the end of R_1 connected to B—; this end of resistor R_1 is, therefore, more negative than the other end.

This voltage is between the grid and cathode, because the grid return from the signal source connects to the negative end of R_1 , while the postive end of R_1 is connected to the cathode. In other words, the d.c. voltage across R_1 makes the grid more negative than the cathode. This is exactly what a C battery or other bias supply does.

Effectively, with this circuit, the tube biases itself—since its own d.c. plate current furnishes the bias voltage, which in turn sets the plate current! That's why such an arrangement is called *self-bias*. Service hint: the proper value of resistor R_1 is rather easy to determine, as the normal d.c. plate current and the required grid bias voltage are known from the tube chart characteristics. By dividing the grid bias voltage by the plate current (in amperes), we get the value of R_1 .

▶ If we did not use by-pass condenser C_1 , there would be an a.c. voltage drop across R_1 just as there was across R_8 in Fig. 15. (The equivalent circuit in Fig. 17B shows this more clearly.) Such a drop is undesirable both because it will reduce the output across R_L and because it will be introduced into the grid circuit (since the voltage across R_1 acts on the grid). Therefore, R_1 is usually by-passed by a condenser such as C_1 in Fig. 17A.

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If by-pass condenser C_1 open-circuits, the stage gain will be reduced. If it short-circuits, the bias voltage will be removed; this will cause a high d.c. plate current and operation on the wrong part of the characteristic curve. Compare this with what you just learned about the effects of an open or a short in the by-pass condenser across the plate supply. Notice — although the two condensers serve much the same purposes, defects in them may have quite different effects on the set.

Typical Amplifying Circuits

While all amplifiers are basically alike, there are differences between them — differences in the methods of coupling into the grid circuit, in the signal source used in the grid circuit, and in the types of plate loads. The exact circuit depends upon the use to which the tube is put; whether, for example, it is to amplify low-frequency intelligence signals or high-frequency

carrier signals, or is to be used in television or in sound systems. Suppose we now briefly introduce you to some typical amplifiers, which you will study fully in your next few lessons.

RESISTANCE-CAPACITANCE COUPLING

Fig. 18 shows an amplifier circuit that is frequently found in audio or

video frequency amplifiers (intelligence signal amplifiers similar to E or N of Fig. 1).

Resistors and condensers are used to couple (connect) this stage to the stages before it and after it, so it is called a "resistance-capacitance coupled amplifier." As you can see, the d.c. plate voltage is obtained from the B supply in the usual way. The stage is self-biased by resistor R_2 . By-pass condensers C_2 and C_3 have already been explained.

When an input signal is applied to terminals 1 and 2, the a.c. portion of this signal will pass through condenser C_1 and develop a voltage across resistor R_1 . If there is also a d.c. voltage between terminals 1 and 2, condenser C_1 will prevent this d.c. component from entering the stage. Thus, C_1 and R_1 form a filter, blocking any d.c. voltage which may be between terminals 1 and 2, but passing on the a.c. signal. (The d.c. blocking is necessary to prevent interference with the bias-the d.c. voltage from the previous stage could force the grid highly negative or highly positive, and thus cause operation at some undesirable point on the tube characteristic.)

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The varying a.c. signal voltage across R_1 is alternately added to and subtracted from the bias voltage produced by R_2 . This produces an a.c. variation in plate current, which in turn produces an amplified copy of the signal voltage across the load resistor R_3 .

The voltage across R_3 is passed through another filter, C_4 - R_4 , which passes on only the a.c. signal and blocks the d.c. voltage in the plate circuit from the following stage.

Notice that condensers C_1 and C_4 serve to isolate the stage. On the input side, C_1 prevents any external d.c. voltage from interfering with the grid circuit and, on the output side, C_4 prevents the plate operating voltage from causing an undesirable action in any following stage or device.

▶ In case you're not sure how an a.c. voltage gets across R_4 from R_3 , notice the ground connections. You can trace a complete circuit (for a.c. only) from R_3 to C_4 , C_4 to R_4 , R_4 to ground, ground to C_3 , and from C_3 back to R_3 .



FIG. 18. This basic circuit is the most commonly used type of audio emplifier.

Thus, any a.c. voltage appearing across R_3 is applied to R_4 through condensers C_3 and C_4 . In effect, these condensers place R_4 in parallel with R_3 insofar as a.c. is concerned.

Very many radio circuits are completed in this manner through ground connections, so remember—when you are tracing a circuit, don't stop when you meet a ground connection; trace from the other end of the source until you come to ground again. You can consider that all grounded points are connected just as if wires were run between them.

▶ However, just because ground points are connected together, don't make the mistake of thinking of currents as wandering about aimlessly. Currents only flow in complete circuits because of the voltages in those circuits. Thus, in Fig. 18, the current flow from R_4 will be through C_3 ; it will not flow through R_1 as there is no path from the grid end of R_1 to the B + end of R_3 .

LOW-FREQUENCY (AUDIO) TRANSFORMER COUPLING

Another amplifying circuit is shown in Fig. 19. Here, resistor R_1 furnishes the self-bias and C_1 is the by-pass condenser preventing any appreciable signal voltage drop across this resistor. Condenser C_2 by-passes the B supply.

The grid circuit is completed through the secondary S of transformer T_1 , while the plate circuit of the tube is completed through the primary P of transformer T_2 .

When an a.c. signal is applied to terminals 1 and 2, the resulting current flow through primary P of transformer T_1 sets up a varying flux which induces an a.c. voltage in the secondary S of



FIG. 19. Transformer coupling is employed when high gain is desired, as you will learn later.

this transformer. This a.c. voltage is between the grid and cathode of the tube, and produces a varying plate current.

The plate current variation must pass through the primary P of transformer T_2 . This produces a flux change in this transformer, and so induces a voltage in the secondary S which can be applied to a reproducer system or to another stage.

► Notice the voltage *changes* which take place. One voltage causes a current to flow in the primary of trans-

former T_1 . This current flow reduces a flux change which induces another (but similar) voltage in the secondary winding. This acts as the input signal for the tube, which varies the plate current of the tube. In turn, this plate current flow through the primary of transformer T_2 causes a flux change which induces a *final voltage* in the secondary S of this transformer. Thus, while the original voltage applied to terminals 1 and 2 gets no farther than the primary of transformer T_1 , successive "carbon copies" are produced by each transformer and by the tube: finally, an amplified copy of the original signal can be obtained from terminals 3 and 4.

The transformers act as "isolators" like the condensers C_1 and C_4 of Fig. 18, in that d.c. has no influence on their secondary circuits. Basically, therefore, we use d.c. to get proper tube operation, but then we usually separate the signal from the d.c. before passing it along.

R.F. AMPLIFIER STAGES

The low-frequency amplifier shown in Fig. 19 uses laminated iron-core transformers, which cannot be used at frequencies higher than about 15,000 cycles per second. R.F. carrier frequencies are far higher than this, so air-core or powdered-iron core transformers must be used in r.f. amplifiers. A typical circuit for an r.f. amplifier stage is shown in Fig. 20.

It is customary to tune radio frequency transformers so that the amplifier stage will give both amplification and selectivity. Condensers C_1 and C_4 in Fig. 20 are variable condensers which are used to tune L_2 and L_4 (respectively) to some desired radio frequency. Let's see what effect the addition of these condensers has on operation of the circuit. An r.f. voltage applied to terminals 1 and 2 causes a current flow in L_1 which sets up a varying flux. This induces in coil L_2 a voltage which acts as if it were in series with this coil. Hence, the tuned circuit L_2 - C_1 is series resonant (has minimum impedance) for the frequency to which the circuit is tuned.

If the voltage induced in L_2 is of the resonant frequency, it will undergo



FIG. 20. This tuned amplifier circuit is known as a t.r.f. stage.

resonant voltage step-up and appear across C_1 in amplified form^{*}; if it is not of the resonant frequency, it will appear across C_1 , but will not be much amplified by the action of the resonant circuit. Thus, if the signal applied to terminals 1 and 2 consists of a mixture of several frequencies, only the resonant-frequency voltage will appear across C_1 in greatly amplified form (though voltages with frequencies near the resonant frequency may be slightly amplified).

The r.f. voltage developed across C_1 is between the grid and the cathode, so it causes a varying plate current through L_3 , which in turn induces a voltage in L_4 . Again there is a series resonant circuit (L_4-C_4) ; and again the resonant-frequency voltage will appear across the condenser (C_4) in amplified form. This voltage can be passed on to the next stage for further amplification. Thus, this r.f. amplifier has variable gain: it amplifies the resonant frequency signal much more than other signals. We can (and, in fact, do) use such a circuit as a selector as well as an amplifier. When we tune its two resonant circuits to some desired frequency, signals of that frequency will receive far more amplification than will signals of other frequencies which may be present in the input to the amplifier. In other words, the circuit will select signals of the desired frequency.

▶ Fig. 20 looks very much like Fig. 19. However, the use of air-core transformers (instead of laminated ironcore transformers) and the fact that tuning condensers are used tell us that this is an r.f. circuit exclusively; aircore transformers and tuning condensers are not used in a.f. stages.

An interesting variation of this circuit can be had by using C_4 to tune L_3 instead of L_4 . This would form a parallel resonant circuit in the plate circuit which would offer maximum impedance at its resonant frequency but far less impedance at other frequencies. In other words, at the resonant frequency the load resistance is a maximum—which, as you have learned, means that the stage gain will be highest at this particular frequency. As a result, the greatest possible amount of the signal voltage will be

^{*}Remember, the action of a series resonant circuit in stepping up a voltage is as follows: The coil and condenser together have a resonant frequency at which their reactances cancel each other; when the voltage induced in the coil is at this resonant frequency, a large a.c. current flows in the resonant circuit. This a.c. current naturally produces a large a.c. voltage drop across the coil and another across the condenser, as each one *alone* has a high a.c. reactance. The resulting voltage across either the coil or the condenser is therefore many times greater than the original induced voltage.



Here is a photograph of the chassis of a six-tube all-wave superheterodyne. The various stages can be recognized in two ways—by the types of tubes used and by their associated signal sources and loads. (You will learn more about this in other lessons.) In this figure, the following parts are identified: A, 110 volt a.c. line cord; B, power transformer; C, rectifier tube; D, filter condensers; E, antenna coil; F, frequency converter tube; G, oscillator coil; H, tuning condensers; I, first i.f. transformer; J, i.f. amplifier tube; K, second i.f. transformer; L, second detector (called the demodulator) tube; M, first audio tube; N, power amplifier tube; O, loudspeaker cable.

developed across L_3 at the resonant frequency. Again we have a *selective* circuit which provides maximum gain at the resonant frequency, but less gain at other frequencies.

► Naturally, the coil and condenser combination in each of these circuits must be properly chosen so that their L-C values will give resonance at the desired frequency. It is possible to change the frequency to which the circuit is resonant (or, as radio men say, to "tune" the circuit) by varying the tuning condensers C_1 and C_4 in step.

► We have given you this brief introduction to amplifiers to show you how similar they are, even though they are designed to perform different jobs. You are going to study the details of these circuits in the next few lessons.

Tubes as Signal Generators and as Signal Mixers

The amplifying stages we have discussed so far have been purely relaying devices. Each has taken a signal from the signal source, made an enlarged copy of it, and passed it along to another stage. The signal itself has not been changed, except that it has been made larger.

There are stages which perform other functions instead of amplifying, or in addition to amplifying. We now will discuss several of these briefly, so you will see the important parts they play.

THE OSCILLATOR

The radio stage we call an oscillator is the stage in which an r.f. signal is generated. This signal is an a.c. voltage which varies between positive and negative values just like the a.c. voltage in your power line. In fact, the only differences between power line voltages and r.f. voltages are in their voltage values and frequencies.

Essentially, here is how an oscillator works. As you know, an a.c. grid voltage in an amplifier stage causes an a.c. plate current. This current produces an a.c. voltage drop across the load which is an amplified version of the grid signal voltage. To make such an amplifier into an oscillator, all we need do is to feed part or all of this a.c. load voltage back into the grid circuit with the proper phase. Then the a.c. load voltage will act as a grid signal, which in turn will cause more a.c. load voltage, and so on. An oscillator, in other words, is simply an amplifier which supplies its own grid signal from its plate circuit.

A typical oscillator circuit is shown in Fig. 21. Here, the a.c. plate current

flowing through the feedback coil $L_{\rm T}$ produces an a.c. flux. This flux links with L_1 , inducing a corresponding a.c. voltage in it (much as if $L_{\rm T}$ were the primary and L_1 the secondary of a transformer). The grid voltage thus developed is fed to the grid (through the parallel $R_{\rm g}$ - $C_{\rm g}$ combination, the function of which we will discuss shortly) and there produces a continuation of the plate current variations, which in turn keeps the grid voltage changing! Obviously, this action can keep on indefinitely, once it gets started. But how does it start?

Because of the L_1 - C_1 circuit, all we need do is to cause some change in plate current. The act of turning on the circuit (applying operating voltages) is sufficient. The rise in plate current as the tube warms up will cause a pulse to be induced in L_1 from L_T . From here, the resonant circuit takes charge and controls the oscillator.

The initial pulse of energy causes a current flow which charges the condenser C_1 . Then, when no further energy is induced, C_1 discharges through L_1 . Electrons flow from the condenser with a rush at first, then, as the condenser voltage drops, the discharge current decreases. In other



FIG. 21. A basic oscillator circuit.

words, the condenser discharge produces a varying current through the coil. And, as you know, a coil opposes changes in the current flow through it. Therefore, when the condenser tries to discharge through the coil, energy is stored in the magnetic field of the coil. When the electron movement has caused the condenser voltage to reduce to zero, the current flow would tend to stop. However, now, the coil field collapses and keeps the current flowing in the same direction as it was going. This results in the condenser being charged up again, this time with the opposite polarity.



A typical midget receiver. It uses the same stages as a large console type. The main differences are: I, a different power supply; 2, different tubes; 3, components smaller physically.

When the magnetic field is completely collapsed, the condenser will try again to make current flow through the circuit, this time in the opposite direction. Thus, if a pulse of energy is fed into the circuit, an oscillatory (back and forth) electron flow will take place in the L_1-C_1 circuit.

This oscillating current flow keeps reversing the polarity with which condenser C_1 is charged. In fact, it follows a sine-wave pattern, with a frequency which is equal to the resonant frequency of the L_1 - C_1 circuit. Thus, our resonant circuit is able to convert the initial pulse of energy into a sine-wave voltage.

If there were no resistance in the circuit, current oscillations would keep on indefinitely. However, any practical circuit always has resistance in the coil and leads. Current flow through this resistance produces power losses (I²R) which rapidly reduce the amplitude of the current oscillations.

This is where the tube comes in. The sine-wave voltage across C_1 controls the plate current, which in turn induces pulses of energy from L_T into L_1 at just the right time to keep oscillations going. This feedback makes up for the power lost in the resistance of the resonant circuit, so the oscillations are maintained.

We can make use of the a.c. power produced in the oscillator circuit L_1-C_1 by placing a coil L_s near coil L_1 . A sine-wave voltage will then be induced in L_s , which we can feed to other circuits, where it can be used as an input signal.

Automatic Bias. Perhaps the easiest way to see why $R_{\rm g}$ and $C_{\rm g}$ are used in the grid circuit is to imagine what would happen if they were not present and the C_1 voltage were fed directly to the grid. Then, when the a.c. voltage across the condenser went through its positive alternation, the grid would become highly positive and a large plate current would flow. In fact, under these conditions, a high-power tube could actually pass enough current to melt the tube elements. R_{g} and C_{g} supply a bias voltage which keeps the grid voltage low enough to prevent this action.

When the grid is driven positive by the voltage across C_1 , grid current will flow because the grid will attract some of the electrons emitted by the cathode. This grid current flow develops a voltage across R_g with the polarity shown in Fig. 21. This voltage charges up condenser $C_{\mathbf{s}}$.

When the L_1-C_1 circuit drives the grid negative, the grid electron flow stops, but condenser C_g now partially discharges through R_g . This discharge current replaces the grid current and maintains the voltage across R_g practically constant. Then, on the next positive swing, C_g is recharged, and events repeat themselves. This is called an *automatic bias*. It is another form of self-bias; here the grid current flow produces the bias, while in those circuits we studied earlier the plate current flow produced the bias. (1) as the generator of the carrier wave in a transmitter; (2) as the local oscillator in a superheterodyne receiver; and (3) as a test signal generator used by servicemen. These are all r.f. oscillators. Audio oscillators are used mostly in test and measuring equipment.

► This is by no means the only oscillator circuit. You'll find many others in later lessons. But while these oscillators are different in some respects some to the extent that they produce sawtooth or square waves instead of sine waves—they are alike in being



Many manufacturers furnish complete service information on their receivers in the form of service sheets and manuals. This layout of the top of a chassis is typical of the kind of information you can expect. This layout gives: 1, the positions of the trimmer condensers used for adjusting the circuits to track together; 2, the types of tubes; 3, the functions of the tubes. Thus: 1, a 65K7 r.f. amplifier; 2, a 65A7 first detector and oscillator combined (the frequency converter); 3, a 65K7 i.f. amplifier tube; 5, a 6F6G output tube; 6, a 5Y3G rectifier tube. The positions of the stages are different in each radio, but thera will always be relatively short signal paths from stage to stage.

Oscillator Uses. With the oscillator we've described, we could produce sine-wave voltages of almost any frequency we desire—from audio to radio frequencies—merely by using the proper values of inductance and capacitance in the tuned circuit. The chief radio uses of oscillators are: amplifiers which supply their own grid signals.

MODULATED AMPLIFIERS

Now that we know how a carrier signal can be generated, let us see how we can modulate it with an intelligence signal. (This is accomplished in sec-

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tion \hat{C} in Fig. 1.) At this point in your Course, we won't give a detailed discussion of the circuits used for modulation, but you should learn the essentials of the process. And, the method used determines the kind of demodulator necessary in a receiver. There are two systems now in wide use; we shall describe both briefly.

► In one important method of modulation, the amplitude (the strength) of the carrier is varied by the intelligence signal. This method—called amplitude modulation (or a.m.)—is used in the standard broadcast stations, in the majority of commercial stations, and for the video television signal.

► Another method is to have the intelligence signal vary the frequency of the carrier. This *frequency modulation* (f.m.) is used for the sound portion of a television program, and for broadcasts on the very high frequencies. In particular, it is used for some police radio and other special services where its unique freedom from atmospheric and man-made interference makes frequency modulation highly desirable.

Amplitude Modulation. Fig. 22Arepresents an unmodulated carrier which has a constant amplitude N. We want this carrier to be varied in amplitude by the audio signal shown in Fig. 22B. In other words, we want to produce an r.f. signal like that shown in Fig. 22C, where the amplitude of the carrier varies as the audio signal varies. (Actually there are hundreds of r.f. cycles to each modulation cycle. We can't show them all in this drawing.)

A simple way to get this action is to feed the carrier into the grid circuit of an r.f. amplifier, and to arrange the amplifier so that the intelligence signal can vary its plate voltage. With no intelligence signal, the plate current pulses follow the r.f. variations; the amplitudes of these pulses are fixed by the operating voltages.

When an intelligence signal is applied, its positive alternations will add to the d.c. plate voltage, while its negative alternations subtract from the d.c. voltage. Thus, the plate voltage is varied up and down. You know that the amount of plate current depends on the plate voltage, so you can see that this variation forces the plate current to vary. Thus, the r.f. pulses are made larger and smaller, accord-



FIG. 22. When the carrier at A is modulated by the audio signal, B, the result can be represented by C.

ing to the intelligence signal variations.

Frequency Modulation. In frequency modulation, the amplitude of the carrier is unchanged, but its frequency is varied. This is quite different from amplitude modulation.

The carrier signal is just like the carrier used in an a.m. system, but the "carrier" frequency is known as the *resting* frequency; it is the frequency radiated when there is no modulation.

Suppose we modulate the resting signal with an intelligence signal. Our



FIG. 23. A fundamental representation of a frequency-modulated wave.

f.m. circuit then produces an output which consists of a signal of constant amplitude that varies in frequency. The frequency increases when the intelligence signal cycle is positive, and decreases when it is negative; the amount of increase or decrease from the normal value depends upon the amplitude of the intelligence signal (the greater the amplitude, the greater the frequency change).

Fig. 23 shows this process in a simplified form. The resting frequency is represented at 1. When modulated by an intelligence signal of mediumstrength, the frequency changes up and down as shown at 2, 3, 4, and 5. A stronger intelligence signal causes a wider frequency change, as shown by 6, 7, 8, and 9. (These represent peak values.)

► To sum up: in an f.m. system, the rate at which the carrier frequency changes corresponds to the intelligence frequency, while the *amount* of carrier frequency change depends on the amplitude of the intelligence signal.

► The actual circuits used for frequency modulation contain a number of stages and are rather elaborate. However, their fundamental operation can be shown by Fig. 24.

This is the circuit of an oscillator which generates a carrier signal at a frequency determined by the tuned circuit L_1 - C_1 . Condenser C_1 is attached to a microphone so its capacitance can be varied by movements of the microphone diaphragm. When no sound waves are striking the microphone diaphragm, the condenser has a normal or resting value, and the oscillator will produce an r.f. carrier at a frequency known as the "resting" frequency.

When sound waves strike the microphone M, the diaphragm is moved back and forth. This varies the capacity of C_1 and so varies the resonant frequency of the L_1 - C_1 circuit. This, in turn, varies the output frequency of the oscillator—in other words, gives us a frequency-modulated carrier. This is a crude example, but it does show the essentials of this system of modulation.

FREQUENCY CONVERTERS

The frequency converter, which combines two different radio frequen-



FIG. 24. This simple circuit shows one possible way of frequency-modulating an oscillator. Of course, practical circuits are far more complex.



FIG. 25. In the mixer-first detector stage, the incoming signal is mixed and combined with the local oscillator signal. The result is a new carrier frequency known as the intermediate frequency.

cies to form a third frequency, is a very important section of a superheterodyne receiver. You will recall that this is the section which takes the incoming signal and combines it with a signal from a local oscillator so that they produce an intermediate frequency (i.f.).

This frequency conversion takes place in a mixer-first detector circuit like that shown in Fig. 25. The name of the stage comes from the fact that two r.f. voltages (the incoming carrier and the output of the local oscillator) are fed into the grid circuit, where they *mix*, and from the fact that the C bias voltage is adjusted so it causes plate current cut-off, which makes this a special form of class B amplifier called a *detector*.

If the tube were a perfect amplifier, its plate current would be just a mixture of the signals fed to the grid of the tube. However, by using the characteristics of a detector (to be fully explained elsewhere), we find that the plate current pulsations contain not only the two original signals but also frequencies equal to the sum of, and the difference between, these two. This difference-frequency component is the i.f. we want, so we tune the resonant circuit L_5 - C_4 in Fig. 25 to this difference frequency, thereby making the stage output consist almost solely of i.f. voltage. This same action occurs on both a.m. and f.m. signals, because the modulations are just transferred to the new i.f. carrier.

The Diode in Radio Systems

While all the circuits we have discussed so far use tubes with control grids, diode tubes (which have only a cathode and a plate) are also used widely in radio—particularly as rectifiers in power packs and as detectors for separating the intelligence signal from the i.f. signal. Let's touch briefly on each of these important uses.

POWER PACK CIRCUITS

The power pack of an a.c.-operated radio or television apparatus is used to convert the a.c. line voltage to the various a.c. and d.c. voltages required by the apparatus.

The power pack in the average superheterodyne receiver (as well as in the average transmitter) has four important components:

1. The power transformer, which changes the a.c. line voltage to a higher a.c. value for the rectifier tube, and to lower a.c. values for the filaments of all tubes in the apparatus.

2. The rectifier tube, which converts the stepped-up a.c. voltage into a pulsating d.c. voltage. 5. The filter section, which smooths out or filters the variations in the pulsating d.c. voltage.

4. The voltage divider, which divides the resulting d.c. voltage into the varicus values required by the grids and plates of individual tubes.

► A basic power pack, similar to those used in superheterodyne receivers, is shown in Fig. 26. Let us learn a little more about the purpose of each part. windings are used for each voltage.

Thus, when 110 volts a.c. are supplied to primary P_1 in Fig. 26, secondary S_1 provides 440 volts a.c.; secondary S_2 provides 5 volts a.c. for the diode rectifier tube filament; secondary S_3 provides 6.3 volts a.c. across terminals x and y for the filaments of all other tubes in the receiver.

The Rectifier Tube. The simplest and easiest way of converting a.c. into



FIG. 26. A simple power supply using a diode tube as a rectifier.

(There is a complete lesson on power supplies later in your Course.)

The Power Transformer. The voltage induced in each secondary winding will depend upon the number of turns in the winding. If the secondary has more turns than the primary, its voltage will be higher than the primary voltage; if the secondary has fewer turns than the primary, its voltage will be lower. If different voltages are required, separate secondary pulsating d.c. is by means of a diode, as you learned in a previous lesson. This tube allows electrons to flow from its cathode K to its plate P when the a.c. voltage makes the plate positive, but completely blocks the flow of electrons when the a.c. voltage reverses polarity. Thus, if the voltage across P_1 is represented by graph A and the voltage across S_1 is represented by B, the current through the diode tube is represented by C.

The Filter Section. The plate current flowing through the diode rectifier tube is varying or pulsating: as you have learned, this means that it contains a d.c. component plus one or more a.c. components. Since we want only the d.c. for our tube plates and grids, we use a low-pass filter to remove the a.c. components. The chokes offer little opposition to d.c., but L_1 - C_2 and L_2 - C_3 form voltage dividers for a.c., in which most of the a.c. is dropped across the high-reactance chokes rather than across the low-reactance condensers. The exact manner of filtering, and the special action of C_1 will be covered fully in later lessons on power supplies. However, notice that graphs D, E, and F show that the result is a progressive reduction in a.c., leaving a relatively pure d.c. voltage across C_3 .

The Voltage Divider. The output voltage at C_3 may be too high for many of the tubes, so a resistor, having taps or terminals along it, is sometimes connected across the output of the filter to divide the d.c. voltage. The arrangement is shown in Fig. 26 and the voltage division is represented at G. It is customary, when a power pack has several output voltages, to mark B_+ on the circuit diagram at the terminal which has the lowest positive voltage; the next higher voltage is indicated by B_++ , the next by B_+++ , and so on.

AMPLITUDE DEMODULATORS

A typical a.m. demodulator stage is shown in Fig. 27A. This corresponds to the demodulator stage M in Fig. 1, and is the place where the intelligence signal is separated from the i.f. carrier. (You will recall that the intelligence signal was transferred from the original r.f. carrier to the i.f. carrier in the mixer-first detector stage.)

In an amplitude-modulated system,

the modulated i.f. fed into the demodulator stage has a form like that in Fig. 27B. The envelope of the high-frequency voltage — that is, the dotted lines drawn through the peaks of the i.f. voltage—represents the intelligence signal we want. There are actually two envelopes in this diagram, one drawn through the positive peaks and the other through the negative peaks. We want either one but not



FIG. 27. The diode is also commonly used as a demodulator.

both, since the two are just opposite. If we attempted to send both into some reproducer they would cancel one another and produce no net effect.

Therefore, we apply the modulated i.f. to the diode circuit shown in Fig. 27A. The diode does not permit current flow during the negative alternations of the voltage, so the output of the tube consists of only half of the modulated i.f. (Fig. 27C).

Remember, now—what we actually want is just the *peak voltage* of these half-cycle pulses. Therefore, we feed the diode output into condenser C_2 . During each half-cycle that the tube conducts, condenser C_2 charges rapidly to the full peak voltage of the pulse; during the rest of the cycle, when the tube is not conducting, C_2 discharges through resistor R_1 . However, R_1 is made so large that C_2 is able to discharge only a small amount before the next pulse comes along and charges it up again. Therefore, the voltage across C_2 (Fig. 27D) remains at or very near the peak voltage of the modulated i.f. pulses all the time. Since C_2 is in parallel with resistor R_1 , the voltage across R_1 (Fig. 27E) also remains just about at the modulation peaks-in other words, the voltage across R_1 is the intelligence signal we want.

FREQUENCY DISCRIMINATORS

The "detector" or "demodulator" for a frequency modulated signal has another function, in addition to that just described. As you learned earlier, frequency modulation changes the amplitude variations of the original intelligence signal into a frequency change. This produces a signal which varies in frequency but has a constant amplitude (Fig. 23). We must convert the frequency variation back into an amplitude change before we can reclaim the original intelligence signal. The processes of conversion and demodulation both occur in a single stage known as a frequency discriminator.

Before we study the circuit, let's review one more fact about resonant circuits.

Resonance Curves. You know that an L-C resonant circuit responds best to a signal that has the frequency to which it is resonant—in fact, it actually "amplifies" or steps up such a signal. However, this does *not* mean that a resonant circuit excludes signals of other frequencies. It responds best to the frequency to which it is tuned, but it will respond somewhat to nearby frequencies, with its response becoming less and less the farther away from resonance we go.

This fact is shown by what is called a "resonance curve," which shows the response of the circuit to the resonant frequency and to other nearby frequencies. A typical curve is showr in Fig. 28. The scale at the left represents the response of a circuit which is tuned to 100 kc. This is indicated by the fact that this is the highest point on the curve. In this case, 100 kc. signals would receive a step-up of 30. Other frequencies are also stepped up, though not as much: frequencies of 90 and 110 kc. are stepped up by a factor of 15, while frequencies of 80 and 120 kc. are stepped up only about 3 each.

Let's see how we can use this variable step-up effect of a resonant circuit



FIG. 28. The response curve of a resonant circuit.

to convert frequency variations into amplitude variations.

Frequency Discrimination. One kind of frequency discriminator circuit is shown in Fig. 29A. The circuit L_1 - C_1 is tuned to the i.f. "carrier" or resting signal, but responds well to frequencies on both sides of this value. The input signal induces voltages in both L_2 and L_3 , thus causing electrons to flow through diodes VT_1 and VT_2 .

Since electrons move only from the cathode to the plate, electrons flowing through VT_1 come upward through resistor R_1 , producing across this resistor a voltage drop with the polarity shown. Similarly, electrons flowing through VT_2 move downward through resistor R_2 ; the voltage drop across this resistor is therefore of opposite polarity from that across resistor R_1 .

The resonant circuit M (L_2 - C_2) is



FIG. 29. One type of f.m. discriminator.

tuned to a frequency somewhat lower than L_1-C_1 , while N (L_3-C_3) is tuned to a frequency somewhat higher. If the frequency of 4 mc. in Fig. 29B represents the L_1-C_1 resonance frequency, then 3.9 mc. may represent the resonance frequency of M, and 4.1 mc. that of N. Hence, neither M nor N is tuned to favor the center frequency, although both pass it along to a certain extent.

The circuits are adjusted so that, at the resting frequency, the voltages applied to VT_1 and VT_2 are equal, and equal currents flow through R_1 and R_2 . As these are equal resistances, their voltage drops are equal, and, since their voltage drops are opposite in polarity (see Fig. 29A), there is NO VOLTAGE between terminals X and Y when only the resting frequency is fed into this stage.

▶ Now let us suppose the signal frequency is modulated so that it swings from 4 to 3.9, back through 4 to 4.1, then back toward 3.9, etc.

As the frequency approaches 3.9 more voltage will be developed by M. This causes larger voltages to be applied to VT_1 ; larger currents result and produce larger voltages across R_1 . At the same time, the signal is moving away from the resonant frequency of N, so less voltage is being applied to VT_2 . The resulting smaller current produces less voltage across R_2 . Hence, for this swing, the voltage across R_1 increases while that across R_2 decreases. The two voltages no longer cancel one another, so a net voltage (equal to the difference between them) appears between terminals X and Y.

For example, suppose the R_1 and R_2 voltages are initially 50 volts each, and cancel exactly. On the swing just described, the R_1 voltage may go up to 90 volts and the R_2 voltage may go down to 10 volts. The difference is 90 minus 10 or 80 volts, which will appear between X and Y. Terminal X will be positive, as the polarity will be that of the voltage across resistor R_1 .

▶ When the swing reverses toward 4.1 mc. exactly the reverse action occurs. Now, N develops the larger voltage, VT_2 passes the larger current, and the drop across R_2 is larger than that across R_1 . This reverses the polarity of the difference voltage existing across X and Y, since the polarity now is that of the voltage across R_2 . Summary: A signal, shifting in frequency, is introduced into this stage. The frequency variations produce a varying voltage across terminals X and Y. Small frequency changes produce small voltages, as frequencies near 4 mc. do not "climb" as high on the L_2 - C_2 and L_3 - C_8 resonance curves. Larger changes, out to the limits set by the resonance points (3.9 and 4.1 in Fig. 29B), produce larger voltages. Thus, we have a circuit which produces an amplitude variation from a varying frequency signal.

This same circuit also gives us de-

modulation; in this, it is like the diode detector we just discussed for amplitude-modulated systems. The diodes VT_1 and VT_2 rectify the r.f. signal, while condensers C_4 and C_5 reproduce the modulation peaks. Thus, the original intelligence signal appears across terminals X and Y.

► Remember, we don't expect you to grasp all the details of operation of these circuits from the brief descriptions we've given! This has been just a preview of what you'll be studying in your coming lessons—we are introducing you to practical radio stages.

A Complete Superheterodyne Receiver Circuit

To show you how the stages you have studied are used in a modern radio receiver, we have sketched a basic amplitude-modulation type superheterodyne circuit in Fig. 30. (This same circuit would do for f.m. if the second detector were changed to a discriminator circuit and the resonant circuits were arranged for the proper frequencies. A television receiver is also similar, but there would be differences in the tuned circuits and in the output stage, as well as in the reproducer.) Let's go through its operation briefly, as a summary of what you've learned in this lesson.

The antenna system intercepts many different modulated r.f. signals, causing modulated r.f. currents to flow in coil L_1 . A number of signals are induced in coil L_2 , but only the signal at the resonant frequency of L_2 - C_2 will undergo the full resonance step-up.

Thus, the resonant circuit serves as a preselector which partially separates the desired signal from all others. The desired signal and such others as get through this resonant circuit then are applied to tube VT_1 . This tube is a class A amplifier, and its plate current variations through coil L_3 induce a voltage in L_4 . The whole stage acts as a combination preselector and r.f. amplifier.

Resonant circuit L_4 - C_4 is tuned to the same frequency as L_2 - C_2 and serves to separate signals further. Again resonance step-up occurs in this circuit at the desired frequency. (While other frequencies may be getting through, they are not being stepped up as much and so are rapidly being "swamped" or drowned out by the desired signal.)

At the same time, tube VT_2 in an oscillator circuit is producing a frequency which is higher in value than the incoming carrier frequency. The oscillator tuning condenser C_{11} , which is operated by the same shaft as tuning condensers C_2 and C_4 , is adjusted so that the oscillator frequency always

differs from the desired incoming signal by a chosen amount—the i.f. value. The coil L_0 provides the necessary feed-back action in this oscillatory circuit. Then, coil L_{10} couples the oscillating circuit L_{11} - C_{11} to the input or grid-cathode circuit of tube VT_3 .

The oscillator and desired input signals are mixed in this mixer-first detector stage, producing the modulated i.f. signal (as well as signals of various other frequencies) in the plate circuit of VT_3 . The resonant circuit L_5 - C_5 is tuned to the intermediate frequency,



Courtesy Philes

A typical portable receiver. Once again the standard superheterodyne circuit is used, this time with battery-operated tubes.

and so maximum voltage is induced in coil L_6 at this frequency.

Resonant circuit L_6 - C_6 is also tuned to the i.f. value. It assists in separating this particular intermediate frequency from among any others that may be produced by the mixer-first detector tube.

The i.f. amplifier tube VT_4 is a voltage amplifier, delivering maximum signals to the parallel resonant circuit L_7-C_7 , which is also tuned to the i.f. value; thus, maximum i.f. signals are transferred to L_8-C_8 .

By now, the combined selective effects of all these circuits should have made the desired signal so much larger than any other signals that little or no interference can occur. We get most of our selectivity in the i.f. amplifier; earlier selector circuits are used principally to keep down certain kinds of interference caused by signals far removed from the one desired.

The signal is now applied to the second detector; here it is rectified by diode VT_5 , which, together with condenser C_{14} , causes the intelligence signal to appear across resistor R_4 .

The intelligence signal variations across R_4 are then passed through C_{23} , appearing across resistor R_5 in the grid circuit of VT_6 . This voltage amplifier then develops an amplified intelligence signal across R_7 , which is transferred to the grid of VT_7 through C_{24} . This last tube is arranged to deliver power to the reproducer, which is a loudspeaker in this diagram. (Of course, if this were a television set, the reproducer would be a television imagereconstructor tube.)

A power supply circuit using tube VT_8 is also shown in this diagram. As you see, this power supply is exactly like the one previously discussed in this lesson.

Fig. 30 thus shows all the basic circuits you will find in a superheterodyne receiver intended for an amplitude modulated signal. Of course, this is a simplified diagram; actual sets usually have variations from these basic circuits which give certain desired actions. Furthermore, multi-element tubes are generally used in place of tubes VT_1 , VT_3 , and VT_4 , and sometimes in place of tube VT_{7} . (In fact, unless screen grid or pentode tubes are used as VT_1 and VT_4 , these stages would have to be modified to prevent undesired oscillations from occurring



FIG. 30. This is a simplified, basic circuit of a superheterodyne receiver. Of course, the actual circuit would use different tubes and would be somewhat more complex, as you will see later. Also, sometimes the r.f. stage might not be used, and more than one i.f. stage frequently is found. An a.c.-d.c. or battery receiver would use a different power supply. It is important to realize, however, that regardless of the "extras" on the receiver, all superheterodynes can be reduced to a basic circuit like this one. As you read the text, notice how this schematic diagram is drawn to show the orderly progress of the signal from the antenna to the reproducer. However, schematic diagrams are intended to show only the ELECTRICAL connections, not the PHYSICAL positions of parts. Compare this diagram with some of the layout sketches shown elsewhere in this lesson.

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within them.) However, this diagram shows you the relationships between stages and prepares you for the detailed discussions of separate stages that are coming in your next lessons.

REVIEWS AND PREVIEWS

Looking back over the lessons you have studied, you will realize that we have gone over the complete radio system more than once. The first time, in an elementary form, we showed how certain radio *parts*—coils, condensers, resistors, and tubes — were found throughout the radio system.

We studied these parts individually, then in small groups and simple circuits. Finally, in this lesson, we have put these parts together into groups that we call *stages*. Now that you have learned the names and purposes of radio stages, and know the order in which radio signals progress from one to another, you are ready to learn the details of operation of each of them.

 \blacktriangleright Now, let's look at the lessons just ahead. Your next one will be on iron-

core choke coils and iron-core transformers. You have already been introduced to these devices, but they have a number of important characteristics which have much to do with the workings of intelligence frequency amplifiers and power supplies. Once these characteristics are understood, you will plunge right into a study of power supplies, and from there into other stages. Soon, you will have a complete "how and why-it-works" understanding of radio equipment.

Remember, once you understand fully how stages and sections work when *normal*, you will know what to expect when trouble occurs. As you go along, you will see that certain breakdowns produce certain easily recognizable effects. For example, the output of the radio device may be distorted in a special manner, or there may be noise or hum, etc., caused by certain particular breakdowns. This is vital knowledge—the kind that leads you right to the trouble, so you are going to find future lessons more and more interesting and valuable!

THE N. R. I. COURSE PREPARES YOU TO BECOME A RADIOTRICIAN & TELETRICIAN (REDISTERTO U.S. PATENT OFFICE) (REDISTERTO U.S. PATENT OFFICE)

Lesson Questions

Be sure to number your Answer Sheet 10FR-4.

Place your Student Number on every Answer Sheet.

Send in your set of answers for this lesson immediately after you finish them, as instructed in the Study Schedule. This will give you the greatest possible benefit from our speedy personal grading service.

- 1. Briefly state the fundamental principle on which the superheterodyne receiver operates.
- 2. Which one of the following amplifier classes duplicates the *entire* grid, signal in its plate current variations: 1, class A; 2, class B; or 3, class C?
- 3. What important tube rating tells *directly* how much the grid controls the plate current?
- 4. When does the plate load in a tube amplifier absorb maximum power?
- 5. Name the three important purposes served by the by-pass condenser C_2 in Fig. 13.
- 6. If the by-pass condenser C_1 in Fig. 17A short-circuits, does the bias voltage: 1, increase; 2, drop to zero; or 3, remain the same?
- 7. In the oscillator circuit shown in Fig. 21, which two parts develop the automatic C bias?
- 8. What are the four important components of an average power pack?
- 9. Which component of the power pack converts a.c. to pulsating d.c.?
- 10. What two processes occur in the frequency discriminator stage of an f.m. receiver?

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ENTHUSIASM

Starting work in a new field of endeavor just naturally arouses an intense and eager interest in what you are doing—an enthusiasm which not only makes study and work a pleasure, but also betters the chances for success.

My students have enthusiasm for their Course of training because it is preparing them for a definite goal—an independent business or a good job. They are continually finding, in the lessons studied, explanations for mysteries which they have encountered. Discovering immediate uses for fundamental facts keeps their enthusiasm high.

When enthusiasm is aroused in an *ambitious* man, he reacts like a thoroughbred race horse, giving all the speed and effort in him. If you try to arouse enthusiasm in a mule, however, all he will do is kick!

A real and lasting enthusiasm for radio will make your study and work as pleasant as play, and will make your life much happier.

J. E. SMITH





STUDY SCHEDULE NO. 11

For each study step, read the assigned pages first at your usual speed. Reread slowly one or more times. Finish with one quick reading to fix the important facts firmly in your mind, then answer the Lesson Questions for that step. Study each other step in this same way.

A transformer transfors POWER from one circuit to another. This is a very important fact—it explains why the currents and voltages have the definite relationship you will find described. Be sure to study carefully the practical section on the power transformer. Answer Lesson Question 6.

This audio transformer is used for voltage amplification, as well as a coupling device between stages. However, as you will see, fidelity requirements are such that a transformer gain of 3-to-1 is considered good. Answer Lesson Question 7.

Transformers handling appreciable amounts of power must be designed for efficiency. Hence, they are designed to match impedances so that a maximum power transfer is possible. In receivers, the output transformer used to match the speaker to the power output tube is the best example. Answer Lesson Question 8.

section on identifying transformers and choke coils. Answer Lesson Questions 9 and 10.

- 7. Mail your Answers for this Lesson to N.R.I. for Grading.
- 3. Start Studying the Next Lesson

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HOW IRON-CORE COILS AND TRANSFORMERS OPERATE IN RADIO CIRCUITS

Magnetic Circuits

A^T the low frequencies handled in power supplies and in audio amplifiers, coils and transformers must have high inductances to offer the necessary reactance values. The only practical way of getting such high inductance values is to use cores of magnetic materials, such as iron or alloy steel. (These cores are all commonly known as "iron cores.")

However, the use of an iron core causes inductance devices to have special characteristics that have much to do with the operation of stages in which they are used. For example, the iron cores of audio transformers limit the frequency response or fidelity of audio amplifiers; the iron cores of power transformers prevent them from transferring power with equal efficiency when operated from power lines of different frequencies. It is, therefore, necessary for us to understand the characteristics of these parts so that we may understand the radio stages in which they are used.

► In this lesson, we shall first review a few of the fundamentals about inductances that you learned in earlier lessons. Then, we will consider the general characteristics of magnetic circuits in an iron core. Finally, we will discuss choke coils, power transformers, and audio transformers in detail. In so doing, we will prepare the way for your study of low-frequency stages and power supplies.

MAGNETIC CIRCUIT FACTS

Magnetic circuits have certain things in common with electric circuits, as you learned in an earlier lesson. A simple electric circuit is shown in Fig. 1A: it consists of a source of electromotive force, and a resistance R which opposes the flow of current through the electric circuit. Voltage E and resistance R together determine the amount of current that will flow. A magnetic circuit likewise has a source, which produces magnetic



FIG. I. A comparison between electrical and magnetic circuits proves their actions to be quite similar.
flux, and also has an opposition to the flux. The source of flux is called the magnetomotive force (abbreviated m.m.f.), while the opposition effect of the magnetic circuit is called its <u>reluc</u>. tance._

One example of a magnetic circuit is shown in Fig. 1B. Here the electric current I, flowing through a coil of Nturns, creates the magnetomotive force that produces the magnetic flux; the iron core in series with the air gap forms the magnetic circuit that offers a certain opposition or reluctance to Actually, the this magnetic flux. greatest part of the reluctance occurs in the air gap, for iron offers little opposition to magnetic flux. Thus, the iron acts as a "conductor" for the magnetic lines, just as the connecting wires in Fig. 1A act as electrical conductors.

Another example of a magnetic circuit is given in Fig. 1C. The magnetomotive force is produced here by a permanent magnet, and the controlling reluctance in the magnetic circuit is that offered by the two air gaps associated with the cylindrical iron core. Incidentally, this magnetic circuit is typical of those found in directcurrent meters.

Thus, you can see that electromotive force (voltage) in an electric circuit is comparable to magnetomotive force in a magnetic circuit, resistance is comparable to reluctance, and current is comparable to magnetic flux.

The same general circuit rules are followed also: increasing the magnetomotive force increases the magnetic flux; decreasing the magnetomotive force decreases the flux. Increasing the *reluctance* of the magnetic circuit by changing some part of its path decreases the magnetic flux; decreasing the reluctance of a magnetic circuit

permits the flux to increase.

How is Magnetic Flux Measured? The amount of magnetic flux in a magnetic circuit cannot be measured in the direct fashion we use to measure current in electric circuits. Instruments have been devised that will measure flux, but they are not simple to use; further, a great many measurements usually must be made to discover the total flux flowing, since a single measurement shows only the amount in a small cross-section of the circuit. Fortunately, you need not worry about flux measurements in practical radio work; that is something we can leave to the research scientist. However, the units in which flux and the two other magnetic quantities-magnetomotive force and reluctance-are measured are of interest to us, for we shall meet them constantly in discussions of magnetic circuits. Let's take a few moments to see what these units are.

Units of Force. You know that an increase either of the number of turns in a coil or of the current through the coil will increase the flux-producing ability (magnetomotive force) of the coil. This magnetomotive force therefore can be expressed in terms of ampere-turns.

The ampere-turn is the practical unit of magnetomotive force for coils, as it provides a direct means of measuring the magnetomotive force. However, this unit cannot be applied to permanent magnets. For this reason, seientists use the *Gilbert*, a unit of magnetomotive force that applies directly to permanent magnets as well as to coils. To change ampere-turns to gilberts, simply multiply the number of ampereturns by 1.26; to change gilberts to ampere-turns, multiply the m.m.f. in gilberts by 0.80.

► The magnetomotive force is the

total force acting throughout the magnetic circuit. To express the force developed over a portion of the length of the path, a term giving the magnetomotive force per unit length is also used. This may be given as gilberts per centimeter. This unit of measurement is called the magnetic force, or magnetizing force, to distinguish it from the magnetomotive force.

Units of Flux. The magnetic line of force (used to represent flux) is a unit originally conceived by the great scientist Faraday, and it is retained to this day because of its convenience. The technical name for a magnetic line of force is the maxwell. A larger unit, the kilomaxwell, is equal to 1000 lines of force or to one kiloline.

Flux density indicates how many lines of flux pass through a given unit of area. Flux density therefore can be expressed in terms of lines per square inch, maxwells per square inch, kilolines per square inch or kilomaxwells per square inch. If the unit of area is one square centimeter (metric units) another unit is used; a flux density of one line per square centimeter is known as a gauss. A kilogauss is equal to 1000 gausses or 1000 lines of flux per square centimeter.

Units of Reluctance. Now we can define the magnetic opposition to the existence of flux in a magnetic circuit. If a circuit has a magnetomotive force of one gilbert and this sets up a flux of one maxwell (one line of force), then scientists say that the magnetic circuit has a reluctance of one *oersted*.

Let us see what controls the reluctance of a magnetic path. Turning back once more to electric circuits for comparison, you will remember that the resistance of the circuit can be increased either by increasing the length of wire path, by decreasing the cross-

sectional area of the wire path, or by making the current flow through wire that has a high resistivity, such as nichrome wire. Similarly, in a magnetic circuit, the reluctance of the path for flux is increased by increasing the length of the path, by decreasing the cross-sectional area of the path, or by using materials for the path that have higher reluctivity (higher reluctance per unit volume). Under normal operating conditions, air has much more reluctivity than iron or steel. Different types of iron. vary greatly in reluctivity; cast iron. is very high and wrought iron is low. ► The comparison to electrical circuits is so close that Kirchhoff's laws for magnetic circuits are much like those for electrical circuits. These laws are:

1. All the flux flowing to a point in a magnetic circuit equals all the flux flowing away from that point.

2. The sum of the reluctance drops . in a magnetic circuit must equal the sum of the magnetomotive forces acting in that circuit.

Permeance of Magnetic Circuits. In electrical circuits, we express the ability of the circuit to conduct current as its conductance. This rating is expressed in mhos, and is obtained by dividing the number one by the circuit resistance in ohms. Similarly, the ability of a magnetic circuit to allow magnetic lines to exist is called its permeance. The number one divided by the reluctance in oersteds gives the permeance.

MAGNETIC SATURATION

You know that doubling the source voltage will double the current in an *electric* circuit that contains only resistance. However, you will recall that saturation occurs in a tube, which prevents the current from being exactly proportional to the voltage when a tube is used. Similarly, it is not always true that doubling the magnetomotive force in a magnetic circuit will double the magnetic flux. Except for air and certain non-magnetic metals, the reluctance of a magnetic path varies with the magnetomotive force acting upon it.

The two curves in Fig. 2 illustrate the difference in action of electric and magnetic circuits. Fig. 2A shows the relation between the voltage applied to a resistor and the current that flows through it. As you would expect, the resultant graph is a straight line.

Fig. 2B is the curve for a magnetic circuit containing iron. This curve shows the relation between the magnetomotive force (m.m.f.) applied to a reluctance and the magnetic flux that flows through it. (The magnetomotive force, the reluctance, and the flux correspond, respectively, to the



FIG. 2. These curves show how a magnetic circuit differs from a single resistive circuit in that saturation occurs in the magnetic circuit.

voltage, the resistance, and the current in the electric circuit.) Notice the graph is not a straight line. Instead, it is a curve that gradually becomes horizontal as the magnetomotive force is increased. In other words,

in an iron-core circuit, increases in m.m.f. beyond the point of greatest curvature do not produce proportionate increases in flux—in fact, when the horizontal part of the curve is reached, the magnetomotive force can be in-



FIG. 3. How the tiny magnets line up in a magnetic material and add their forces to the total applied.

creased greatly without producing more than a small increase in flux.

This condition, in which an increase of the magnetomotive force in a circuit produces little change in the flux, is called "saturation," and is similar to the saturation action that occurs in tubes. (You will recall that an increase of the filament temperature beyond a certain point does not increase the emission.) Magnetic saturation occurs only in magnetic circuits that contain *magnetic* materials, such as iron or steel; an air-core coil cannot be saturated.

Scientists explain the phenomenon of magnetic saturation in a very simple way. They say that iron, or any other magnetic material, is made up of millions of tiny individual magnets, which ordinarily are arranged in a random fashion throughout the material as shown in Fig. 3A. When a magnetomotive force is applied to the circuit containing this magnetic material, some of these little atomic

magnets line up parallel to the magnetic lines of force (Fig. 3B), with their North poles all pointing in the same direction. The flux in the circuit then consists of the flux caused by the applied m.m.f., plus the much greater flux caused by the individual magnets; the total circuit flux, therefore, increases tremendously over what it would be in an air-core coil for the same amount of magnetomotive force. If we keep increasing the amount of m.m.f. applied to the circuit, more and more of the individual magnets line up, and the flux continues to increase rapidly. Soon, however, all the individual magnets are lined up (Fig. 3C); they have then added as much to the total flux in the circuit as they can. Then, any further increase in the applied m.m.f. adds to the total flux merely the relatively small amount that an equivalent increase in m.m.f. would add in air. The curve of flux plotted against m.m.f. then becomes almost flat, and we say that saturation has been reached.

▶ The m.m.f. that must be applied to cause saturation depends on the size, material, and construction of the core. Very often, when we are using an iron-core choke or transformer, we do not want saturation to occur when a normal amount of current flows through the coil. For this reason, transformers and choke coils are often made with air gaps in their cores. An air gap increases the reluctance of the core, so a relatively large current must flow through the coil to cause saturation. (We will say more about air gaps later in this lesson.) Large cores also help prevent saturation, for the larger the core, the more flux it can carry before saturation occurs.

B-H CURVES

Engineers use graphs-known as

"B-H curves"—nuch like those in Fig. 4 to show the magnetic characteristics of various materials. These curves are not exactly the same as the curve in Fig. 2B, because the units are different. Flux density—that is, the number of flux lines per square centimeter of the material—is plotted along the vertical scale, and magnetizing force (m.m.f. per unit length) is plotted along the horizontal scale. Flux density is assigned the symbol B while magnetizing force is given the symbol H.

In these curves, B and H are used



FIG. 4. B-H curves are useful to engineers because they show how various materials differ in their characteristics.

instead of flux and m.m.f. so that the information can be made to apply to cores of any size and also so that the curves will apply only to a particular material (not to an entire magnetic circuit, which may contain an air gap or other materials). These curves make it possible to find the characteristics of the material itself in such a way that the information can be applied to any core using that material, whereas the curve shown in Fig. 2 would apply only to a particular core or magnetic circuit. (B and H are also commonly expressed in English units-flux per square inch and m.m.f.

per cubic inch—as well as in the metric (centimeter) units used in Fig. 4.).

Notice that for air the abscissa (horizontal or H scale) readings must be multiplied by 200. Thus, to get a B reading of 5, there must be 5000 Hunits (25×200) for air; this is because air is so much poorer a magnetic conducting material than either iron or steel. The curve for air applies also to non-magnetic materials like brass, copper, wood, bakelite, and fiber. Furthermore, this graph shows that flux density increases uniformly with the magnetic force only for air and non-magnetic materials; all magnetic materials become saturated when subjected to large magnetizing forces. \blacktriangleright Very likely you will never use B-Hcurves in your work, as they are intended primarily for the designer of iron-core devices. They are shown here to familiarize you with them, for you will frequently find them mentioned in technical articles.

IRON CORE LOSSES

So far, we have discussed what might be called "d.c. magnetic circuits"—circuits in which the magnetizing force is supplied by a direct current of steady value flowing through a coil. In radio, however, we are concerned principally with the effects of alternating current flow through coils. As you may suspect, the behavior of coils is considerably more complicated when a.c. is passed through them.

For example, there are certain magnetic losses that every iron-core device has to some extent when a.c. is passed through it—hysteresis (pronounced hiss-ter-E-sis) losses, eddy current losses, and flux leakage losses. (This last loss occurs also in d.c.operated devices, but is much more

important, as you will see, when the device is a.c.-operated.) All of them reduce the efficiency or effectiveness of the device in which they occur, and all can be minimized (but not eliminated altogether) by proper design. As a knowledge of what causes these losses will help you to select low-loss devices for circuits that require their use, let's take a moment to see why these losses occur.

Hysteresis Losses. If we send a current through a coil that has an unmagnetized iron core, then reduce the current to zero, the core of the coil will remain magnetized to a certain extent. (This, in fact, is the way permanent magnets are made.) The core is then said to have "residual magnetism." If we want the core to be completely unmagnetized, we have to do more than reduce the coil current to zero-we must send a small current through the coil in the reverse direction. In other words, we must use up electrical energy to eliminate the residual magnetism of the core.

The reason the core has residual magnetism is that the tiny magnets of which it is composed, once they have been lined up by the magnetic field of the coil current, exert forces on one another that tend to make them stay lined up. Thus, when we reduce the current through the coil to zero, some of these magnets remain lined up and cause some flux to continue to pass through the core. We have to send flux through the core in the reverse direction to turn the magnets from their lined-up positions, thus restoring the core to its original unmagnetized condition.

The property of iron (or other magnetic materials) of retaining magnetism after the magnetizing force is removed is called *hysteresis*. The electrical energy that must be used to bring the iron back to its original nonmagnetic state represents the hysteresis loss.

► Any iron-core device fed with a.c. power will waste part of the power in this manner. As you know, the a.c. current in the coil causes the flux to act first in one direction, then in the other. A certain amount of power is converted into heat in restoring the core to its non-magnetic condition each half cycle, so that the next half cycle of flux change may follow.

We cannot eliminate hysteresis loss altogether, but proper choice of materials will minimize it. For example, hard steel retains its magnetism, link, and the direction in which this induced voltage will act (force a current to flow) is at right angles to the direction of the flux lines.

Now, let's consider the core commonly used in chokes and transformers. For the moment, let's consider the core to be solid, as shown in Fig. 5A. When the flux follows the indicated paths, the iron in the core acts as if it were made up of a great number of parallel rings at right angles to the flux path (that is, across the core—see the drawing). The varying flux in the core induces a voltage in each ring; this voltage sends through the ring a current that is



FIG. 5. Notice the large eddy current paths in the solid core. These are reduced in the laminated core.

which is why it is used for permanent magnets. It is, then, a poor material for a transformer or choke core, as a great deal of power would have to be wasted in overcoming hysteresis. On the other hand, soft iron and annealed silicon steel retain very little of their magnetism, so cores made from these materials would have far less hysteresis loss. These are the materials that are generally used in transformer cores.

Eddy Current Losses. From your fundamental study of inductances, you know that when a varying current flows through a coil, a varying flux is produced in the coil. This flux variation will induce a voltage in any metallic conductor with which it can known as an eddy current.

This eddy current produces two effects. First, it sets up a flux in the opposite direction to the original magnetic flux and thus reduces the amount of change in the total flux through the core. As a result, the inductance of the iron core coil is lowered.

Even more important—you know that iron has considerable electrical resistance. Therefore, the eddy current flowing in each ring produces a power loss (I²R). This power loss is more important than the loss in inductance, because it, together with the hysteresis loss, wastes power that otherwise could be put to useful work. Eddy current losses can be reduced considerably in low-frequency ironcore devices by constructing the core from thin sheets called laminations. as shown in Fig. 5B. (Here the coil is not shown, but it would be in the same position as in Fig. 5A.) These sheets are annealed (softened) in such a way that their surfaces are coated with an iron-oxide scale that is a fairly good insulator. Often the surfaces of the laminations are varnished in addition. to give even better insulation. Insulated in this way, each thin lamination acts like a separate core, and the eddy current rings that form can be no wider than a single lamination. The rings thus become much smaller than in a solid core; less flux can pass through each ring (there is less flux linkage) and therefore less voltage is induced in each ring to cause a power loss.

► Both eddy current and hysteresis losses increase greatly with frequency; this is the reason why even laminated iron cores are of no value for radio frequency purposes. At these frequencies, the only magnetic cores that may be used are made of finely pow-



FIG. 6. Leakage lines of force usually fail to link with the entire winding.

dered iron or steel mixed with a clay or plastic binder. The iron or steel particles are so separated that losses are kept low, yet there is enough of this material in the flux path to reduce the reluctance to the point where the inductance is fairly high.

Flux Leakage Losses. Since iron is a much better magnetic conductor than air, most of the flux produced by a coil will flow through the iron core rather than through the surrounding air. However, as indicated in Fig. 6, a certain amount of the flux always takes the high-reluctance air path. (You may compare this to the action of electrical current through a circuit made up of a high and a low resistance in parallel. Most current will flow through the low resistance, but some will go through the high resistance.) Furthermore, as the core approaches saturation, more flux takes these air leakage paths, for the reluctance of a saturated core goes up very rapidly and soon approaches that of the air path.

Where the flux path through the air links the entire coil winding (or windings, when there are more than one) as shown by line A in Fig. 6, complete flux linkage occurs just as if the flux had passed through the core. However, the other air paths in Fig. 6 do not completely link the coil (or coils). The flux that passes through these paths is called "leakage flux," because it "leaks" out of the desired path. Any leakage from the primary winding of a transformer causes less energy to be transferred to the secondary, because primary leakage flux does not link with the secondary, so it cannot induce a voltage into the secondary.

This leakage flux is undesirable, both because it reduces efficiency and because it can create interference by inducing voltages in nearby circuits. Design engineers minimize it by designing the iron core to operate well below saturation. Often, also, the coil and core are covered with a soft steel casing so leakage flux will flow through this casing and back to the iron core instead of wandering off through the air.

Iron-Core Chokes

As you know, the reactance of a coil depends upon both its inductance and the frequency of the applied voltage. If the frequency is low, the inductance of the coil must be high for it to have a fairly high reactance. When we want a high-reactance choke coil in a low-frequency amplifier or a power supply, we must use an ironcore choke, because this is the only kind of choke that can give us the necessary high inductance and still be reasonable in size.

But, while the use of an iron core in a choke solves the problem of getting high reactance at low frequencies, it introduces other problems. In particular, it introduces the problem of saturation—a very real difficulty, because in most practical choke coil applications, d.c. as well as a.c. flows through the choke coil. The d.c. value is frequently high enough to cause saturation or near-saturation of the choke core.

You learned earlier in this lesson that iron-core choke coils very often have an air gap in the core. Now, such a gap increases the reluctance (decreases the flux) in the core considerably, since air has a very much higher reluctance than magnetic materials. Perhaps it seems strange to you that chokes should be built with iron cores to increase the flux through them, only to have the flux reduced by the addition of the air gap, but there is a good reason.

► As you know, the inductance of a coil depends upon the amount of *change* of flux within the coil, rather than upon the total flux present. (Voltages are induced only by variations in flux.) If the flux change for a given change in current is high, the inductance is high; if the flux change is low, the inductance is low.

▶ We can use the solid-line curve in Fig. 7 to show what may happen when a mixture of d.c. and a.c. flows through a coil that has no air gap.

Let us first assume that a small amount of d.c. flows, and produces the amount of core flux indicated by point I on the curve. Then, let us presume that enough a.c. flows to cause a variation in the flux from point A to point B on the curve.

Let us now draw a horizontal line from point A toward point C, and draw a vertical line from point B to meet it, as shown in Fig. 7. The length of the horizontal line A-C is a measure of the amount of a.c. current change, while the vertical line B-C represents the amount of change in the core flux produced by this current change. The longer the vertical line (B-C) for a particular line length A-C, the greater the amount of flux change for the particular current variation, and hence the greater the inductance.

▶ Now let's suppose that the d.c. cur-



FIG. 7. An air gap reduces the inductance somewhat, but, as a compensation, a magnetic circuit containing an air gap undergoes less change in inductance with d.c. variations; also, a higher current can be applied before saturation occurs. The dotted curve applies to the air-gap core. rent flow is much greater than that for point 1, so that it produces the coil flux represented by point 2, up in the saturation region of the curve. Let's suppose we have the same amount of a.c. The a.c. will now vary the core flux from point D to point \tilde{E} on the curve. That is, if we draw horizontal and vertical lines as before. we find that the line D-F is the same length as the line A-C, so it represents the same amount of a.c. variation. However, when we compare the length of the line E-F with the line B-C, we find that the E-F line is very much shorter. You remember that this vertical line represents the amount of flux change caused by the a.c. Therefore, the increased d.c. flow has, by almost saturating the core, reduced the amount of core flux change that the a.c. can cause. And, since the inductance of the coil depends on the amount of flux change that the a.c. causes, the inductance of the coil has decreased.

► Naturally, we would prefer to operate at point 1, where the coil has its greatest inductance. Unfortunately, the d.c. current in most radio circuits either changes considerably from time to time or is much too large to permit operation at the desired point. Either condition is bad: the first will give us an inductance that varies over a wide range; the second, an inductance that is too low. Therefore, we introduce an air gap into the core to minimize saturation effects. even though we thereby sacrifice some inductance

The air gap, as we said, reduces the amount of flux in the core for a given amount of current. This changes the shape of the current-flux curve. For example, an air gap introduced into the core for which the solid-line curve of Fig. 7 is drawn may produce the

current-flux curve shown as a dotted line in Fig. 7. Notice-this curve is much more nearly linear (straight) than the solid-line curve that represents the current-flux relationship in a solid core This means that we will approach the ideal condition far more closely when we operate along the dotted curve, for equal changes in coil current will produce approximately equal changes in flux as long as the d.c. does not cause operation beyond the straight portion of the curve. Since this curve has a longer straight portion, a much higher value of d.c. is permitted.

Let's use these curves to compare the inductance of the air-gap coil with that of the solid-core coil. Say we send through the air-gap coil the same amount of d.c. that caused operation at point 2 on the solid curve. As you see from Fig. 7, this amount of d.c. produces operation at point 3 on the dotted (air-gap) curve. If we use the same size a.c. current variation (G-I)that we used for operation on the solid curve, we produce a flux variation H-I. This flux variation is considerably greater than the flux variation E-F produced by the same current variation at point 2 on the solid curve. Therefore, our air-gap coil has a greater inductance than the solid-core coil has when this amount of d.c. flows through it.

However, the flux change B-Cproduced in the solid-core coil when operated at point 1 is considerably larger than the flux change produced in the air-gap coil by an equal current change. The air-gap coil, then, has less inductance than the solid-core coil when the latter is operated at point 1. In other words, a solid-core choke may have more or less inductance than a similar choke using an air-gap core, depending upon the amount of d.c. sent through the coil. On the other hand, the air-gap choke will have a fairly constant inductance even with wide variations in the d.c. going through it.

It is quite important that the air gap remain exactly as designed. The space cut into the laminations may be very small. To prevent changes in air-gap spacing, it usually is filled with some non-magnetic material, such as copper or brass strips, cardboard, or paper.

Even with an air gap, it is possible to saturate the core and reduce the inductance of the coil if we allow enough d.c. to flow through it. For this reason, a normal d.c. flow (often called the *polarizing current*) is specified for chokes by the manufacturer. This flow cannot be exceeded much if the rated inductance of the choke is to be secured. Incidentally, if the d.c. flow is less than the specified value, the inductance will be greater than its rated value. When speaking of the inductance of an iron-core coil, always remember that this value is obtained only at a specified current value.

► Since a choke coil is always operated on a.c. or on pulsating d.c. (a combination of a.c. and d.c.), it has all the iron-core losses we have previously discussed. It must be designed to keep hysteresis and eddy current losses low, because it wastes power otherwise. This power produces considerable heat, which may eventually damage the insulation on the coil wire and ruin the choke.

Swinging Choke. While most choke coils are designed with an air gap of sufficient size to avoid saturation within their normal operating current ranges, there is a special type that deliberately uses saturation to *obtain* a change in inductance. Since the inductance varies or "swings" with changes in d.c. flow, it is called a swinging choke.

We won't go into details of the operation of this choke here, as you will find it fully described in later lessons on power supplies. Briefly, it is used as a kind of voltage regulator in power supplies, it changes in inductance (and therefore in reactance) when the current drawn from the power supply varies, and this changed reactance acts to keep the output voltage of the power supply fairly constant.

This choke is made with a very small, carefully chosen air gap, and is designed for a particular range of current values.

Iron-Core Transformer Fundamentals

The ability of a varying magnetic field to induce a voltage in any conductor with which it links makes it possible to transfer power from one circuit to another without direct wire connections. Furthermore, by arranging the linkage properly, we can get a step-up or step-down of voltage and can match impedances.

Suppose we have a coil so placed

that the varying magnetic field produced by another coil links with it. A voltage is induced in the first coil, the amount depending on the flux linkage. This, in turn, depends on the amount of varying flux and on the number of turns of wire on the first coil. (For convenience, the winding connected to the *source* is called the primary, while the other winding, inductively linked to the primary, is called the *secondary*.)

The amount of flux is determined by the design of the primary winding and the primary current. Assuming a fixed current and a particular primary design, the flux linkage in the secondary depends on the number of turns of the secondary winding and on the coupling. (With an iron core, the coupling usually is close to unity, so we can assume perfect coupling for now). The more turns, the greater the flux linkage and the greater the voltage induced in the secondary. We can make the secondary voltage higher or lower than the primary voltage by choosing the proper number of turns for the secondary winding. There is, in fact, a very simple relationship between these two voltages:

The ratio of primary to secondary voltage is the same as the ratio of the turns on these windings.

► Let us take as an example a transformer, having no losses, with 100 turns of wire in its primary. The secondary of this transformer has 300 turns of wire on it. The ratio of turns on the primary to the turns on the secondary is 1-to-3 (100-to-300), so for every volt applied to the primary there will be three volts from the secondary. Thus, the voltage supplied by the secondary will be three times that fed into the primary. If we apply 10 volts to the primary, the If secondary will supply 30 volts. the primary voltage is 110 volts, the secondary voltage will be 330 volts. This is known as a "step-up" transformer, because the transformer steps up, or increases, the voltage.

(In a practical transformer, there would be many more than 100 turns on the primary and 300 turns on the secondary. However, the *ratio* of the turns may be 1 to 3. The actual

number of primary turns depends on the intended use. In selecting the number of turns, the primary voltage and frequency, the type of circuit, the size of the core, and the secondary voltage and current requirements must all be considered.)

▶ This action can be reversed by using fewer turns on the secondary than on the primary. If our transformer has 10 turns on the secondary and 100 turns on the primary, the ratio of primary turns to secondary turns is 10-to-1 (100-to-10). The secondary, having fewer turns than the primary, will have a lower voltage; in fact, the secondary output voltage will be onetenth the voltage applied to the primary. Therefore, if we apply 10 volts to the primary we will get 1 volt from the secondary, while 110 volts fed into the primary will give an 11-volt output from the secondary. Such a transformer is known as a "step-down" transformer, because it steps down, or reduces, the applied voltage.

Thus, a transformer can deliver a higher or a lower voltage from a source to another circuit.

► Here is a practical hint about turns ratios. It is common practice to give the ratio with the larger number first, then to state whether it is a stepup or a step-down ratio. For example, a transformer rated as a 5-to-1 stepup type delivers 5 times the primary voltage from the secondary, while a 5-to-1 step-down transformer delivers 1/5 the primary voltage from its secondary.

POWER TRANSFER

The relationship between primary voltage and secondary voltage is only one of the things you need to know to understand transformer action. We will now show that the relationship between primary and secondary currents is equally as important.

A transformer must always be considered as a *power* transferring device. Hence, the current flow in the primary winding depends on the power taken by the load. We'll use the circuit in Fig. 8 as an example. Here we have a 5-to-1 step-up transformer, which raises the 10-volt source value to 50 volts across R. Let us suppose R has a value of 25 ohms. The secondary current I_s will be 2 amperes $(I_s = E_s \div R)$, and the power dissipated will be 100 watts $(P = I_s \times E_s = 2 \times 50)$.

To supply the load, the transformer primary must draw enough current to take 100 watts from the 10-volt



FIG. 8. A step-up transformer.

source. Hence, the primary current I_P must be 10 amperes $(I_P = P \div E_P = 100 \div 10)$.

Compare the primary voltage and current with the secondary voltage and current. The voltage is steppedup from 10 volts to 50 volts, a step-up ratio of 5-to-1. On the other hand, the current is stepped-down from 10 amps. to 2 amps., which is a stepdown ratio of 1-to-5.

Similarly, when we have a stepdown transformer, the secondary voltage is less than that of the primary, but the secondary current is greater. For example, suppose we apply 110 volts to the primary of a 10-to-1 stepdown transformer; the secondary delivers 11 volts $(110 \div 10)$. Say the secondary power demand is 550 watts; the secondary current will be:

 $I_{\rm s} = P \div E_{\rm s} = 550 \div 11 = 50$ amperes.

Then, the primary current will be:

 $I_{\rm P} = P \div E_{\rm P} = 550 \div 110 = 5$ amperes.

Thus, the voltage goes down from 110 to 11 volts (10-to-1) while the current comes up from 5 to 50 amperes (1-to-10).

► These facts are true because a transformer must transfer power from one circuit to another, and it takes from the source the amount of power required by the load.

The name of the transformer (stepup or step-down) comes from what happens to the voltage—not to the current. Thus: if the secondary voltage is higher than the primary voltage, the transformer is a step-up type; if the secondary voltage is lower than the primary voltage, the transformer is a step-down type. The CURRENT relationships are opposite; that is, in the step-up transformer, the secondary current is less than the primary current, whereas in a step-down transformer. the secondary current is greater than the primary current. As a summary:

If the voltage is stepped up, the current is stepped down; if the voltage is stepped down, the current is stepped up; and the product of the current and the voltage (current multiplied by the voltage) is the same for both the primary and the secondary.

Losses. Of course, we don't actually get a 100% transfer of power. You should keep in mind that there are losses in all transformers caused by eddy currents and hysteresis in the core, which waste power in heat. Also, there are losses in the coils caused by the flow of current through the resistance of the windings. You may see this loss referred to as I^2R loss or "copper" loss. This also takes the form of heat, and varies as the load is varied: the greater the load, the greater the power loss.

► These losses explain why 100% transfer of power from the primary to the secondary cannot be obtained. Actually, the primary source must furnish more power than is taken by the secondary load, because it must supply a total power equal to the load demand plus the losses. As a result, the ratio of primary to secondary currents will not be *exactly* in the ratio of the turns. However, in practice, you can assume the ratios to be nearly correct, as power transfers of 95% to 97% are obtainable in iron-core devices.

Although a 3% to 5% loss seems small, it does limit the amount of power we can transfer through the transformer. The power dissipated in heat increases greatly as the load is increased, so that eventually a point is reached at which the transformer becomes overheated and subject to breakdown. Power transformers are rated according to power-handling ability as well as to voltages and currents delivered. The larger the wire used for the windings (to reduce resistance) and the larger the iron core (to reduce hysteresis and eddy current losses), the more power a transformer can handle before breakdown This is why transformers occurs. having high power ratings are so large and heavy. The weight of a transformer is a good indication of its power-handling ability.

POWER TRANSFORMERS

The power transformer in a radio receiver is an excellent example of both step-up and step-down transformer actions. Homes in this country usually are wired so that power outlets deliver 110 to 115 volts a.c. It is highly desirable to obtain all voltages required for a radio receiver from this power line; this means that we must have low a.c. voltages for the filaments and heaters of the tubes, and high a.c. voltages that can be converted into high d.c. voltages for the various tube electrodes. The line voltage value must be stepped-down (decreased) in one case and stepped-up (increased) in the other case. A single power transformer is used for this purpose.

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While we won't go into the design of a power transformer, it is interesting to know that the engineer considers many factors. Cost, weight, and efficiency are of particular importance. He must choose first a suitable core material, and then must choose a core size that will give him the required characteristics in a minimum of space and weight. There must be a compromise between core characteristics and the number of primary turns. For example, when a transformer is not loaded (nothing is connected to its secondary winding) the flux changes in the core must induce a back e.m.f. in the primary nearly equal to the applied voltage. Otherwise, the primary current (which depends on the difference between the applied voltage and the back e.m.f.) would be excessive. For a given flux density in the core, a given core area, and power line frequency, the back e.m.f. will depend on the turns on the primary.

Therefore, the designer must make a compromise. If he increases the core size, the number of turns in the primary can be made less, but the cost of the transformer, its size, and its weight all increase. Increasing the turns reduces the required core size, but there is a minimum core size too —if the core is too small, saturation may occur before the required flux density is produced.

After selecting a core, the designer computes the number of primary turns that will allow only enough primary current to flow to keep the flux density in the core at the desired value when no load is on the secondary. He then divides this number of turns by the applied line voltmary for each volt applied to the primary winding.

After finding this ratio for the primary, the number of turns required on the secondary winding to secure a certain voltage can be determined by multiplying this desired voltage by the turns-per-volt ratio of the primary. For example, if the primary has a ratio of 10 turns-per-volt, and 500 volts is desired from the secondary winding, the secondary winding



Courtesy Universal Winding Co.

At the left—this view shows the automatic coil-winding machinery used in the manufacture of chokes and transformers. Six or more coils are wound on a single spiralwrapped paper form like one of those shown at the right; when winding is completed and the operator has anchored the lead wires, the machine automatically cuts the coils apart. Operators perform only the starting and finishing operations; the machines do the rest, placing layers of insulating paper between each layer of a winding and stopping automatically when the correct number of turns has been wound. Samples of finished coils produced by machines like these can be seen in the foreground in the righthand phote. The iron core is stacked inside the coil (or coils) to complete the unit. Then, the choke or transformer may be dipped in a moisture-excluding sealing compound and may be enclesed in an iron shield can.

age to find the *turns-per-volt ratio* of the primary. For example, he may find that he has wound 575 turns in his primary winding for a line voltage of 115 volts. Dividing 575 by 115, he finds that he has five turns on the priwill require 500 times 10, or 5000 turns. The turns-per-volt ratio of a power transformer varies considerably with the design of the transformer, since, as we have indicated, it depends on the size of the core, the core material, the power line frequency, and the desired flux density.

Turns Ratio. Although designers of power transformers think in terms of turns-per-volt, you will undoubtedly follow_the practice-of-other transformer users and think of power transformers only in terms of what they can deliver. In other words, you are interested solely in the voltages that may be obtained from each secondary and in the respective current ratings of the secondary windings.

As you know, the output voltage is dependent upon the turns ratio—how many times more turns there are on one winding than on another. The turns ratio times the primary voltage will give you the secondary voltage (provided, of course, you know whether you have a step-up or stepdown transformer action in that particular winding). Remember, there will be a different turns ratio for each secondary delivering a different voltage.

► Should it ever prove necessary, you can measure the turns ratio of a power transformer by connecting it to an a.c. source of known value and measuring the voltage across each winding. The larger voltage divided by the smaller voltage gives you the turns ratio. The winding giving the higher voltage will have the more turns of wire.

► A word of caution—before you connect a power transformer to a voltage source, be sure that the windings will handle that voltage safely, and be sure the frequency of the power source is correct for the power transformer. Otherwise, the current flow may be excessive and the transformer may heat up so much it will be seriously damaged. (You can connect a winding to a source producing a voltage *lower* than its rated value, but never to a voltage source that is considerably higher than its rated voltage.)

Therefore, the first thing you always want to know about a transformer is its primary voltage rating and the frequency for which it is intended. The operating voltages for a power transformer are given by the manufacturer. Most American power transformers are designed for use on 110-115 volt a.c. power lines, though some are designed for 220-230 volt power lines. (You are most apt to find these latter in sets intended for export to foreign countries, where these power line voltages are more common.) Occasionally, too, you may find a transformer especially designed for some unusual power line voltage.

Transformers are usually designed for 25- or 60-cycle operation. Design compromises allow 25-cycle power transformers to operate on 25- and 40-cycle power lines, while 60-cycle transformers will operate on 50- and 60-cycle power lines. However, a 60cycle transformer will overheat if it is connected to a 25-cycle power line, whereas a 25-cycle power transformer will tend to be overly efficient, giving higher-than-rated output voltages, if connected to a 60-cycle power line.

Never connect a power transformer to a D.C. source. On d.c., only the primary resistance limits current flow so the current flow is excessive. The transformer is certain to be damaged or destroyed.

► After making sure the transformer will work properly on the available power line, you will next want to know the voltage and current ratings of the various secondaries. When we take up power supplies, you will learn that a number of different secondary voltages are required. The values needed depend principally on the operating voltages required by the tubes used in the receiver and on the manner in which the tubes are connected (series or parallel connections between filaments).

Each secondary winding will have its own current rating, which is the maximum current that, can be taken from the winding for long periods of time without overheating the transformer. Of course, less current than this can be drawn if desired. Thus, if a filament winding is rated at 2.5 amperes, it can deliver safely any current value up to 2.5 amperes. The current rating needed depends on the number and types of tubes to be supplied.

It is unnecessary to calculate the power handled by a transformer in a receiver as long as you make sure its individual windings have the proper voltage and current ratings. Remember, though, that when a power transtormer has several secondary windings, the primary winding supplies the power required for all of them, plus the power wasted in the eddy current, hysteresis, and I²R (or copper) losses of the transformer itself.

REQUIREMENTS OF AUDIO-FREQUENCY TRANSFORMERS

A power transformer is designed to work at one frequency—that of the power line for which it is intended. A transformer designed to handle audio frequencies must handle a band of frequencies instead of a single frequency, and it is supposed to operate over this band without discrimination —that is, it should deliver a voltage proportional to the turns ratio, regardless of frequency, within its designed range. As you will soon see,

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these requirements are far more difficult to meet than those imposed on power transformers.

Audio Frequencies. The a.c. voltages handled by audio stages correspond to sound waves, and so have the same frequencies as ordinary sound waves. However, although sound frequencies range from 30 evcles per second to about 20,000 cycles per second, good voice reproduction requires only frequencies from about 300 to 3000 cycles, and good musical reproduction requires only frequencies from about 60 to 8000 cycles. Thus, an audio transformer does not have to handle the entire range of sound frequencies. In fact, in ordinary receivers it has to handle only frequencies from about 100 cycles to near 5000 cycles, for cost limitations prevent such receivers from having a wider range. However, even this limited range creates a good many problems. We'll take these up in just a moment, but, before we do, let's see what are the two main types of audio transformers.

► Audio transformers are classified according to whether they are used to give voltage gain or to match im-The types designed for pedances. voltage gain are called interstage transformers, as they are generally used between two vacuum tube stages. Those intended to handle audio power are usually called impedance-matching transformers. In radio receivers, the most common use for such an audio transformer is to connect the power amplifier tube to the loudspeaker. (It is common practice to call this transformer the output transformer.)

Now, let's take up each type in detail and see what problems they will give you in servicing a receiver.

Interstage Audio Transformers

Interstage transformers are normally designed to have a step-up ratio between the primary and secondary windings, so that there will be a gain in signal voltage. These transformers are connected between the source and some voltage-operated circuit that does not require power, so the secondary current flow is negligible. A typical "load" would be a grid circuit, where only voltage is needed for operation. In effect, then, the secondary winding feeds into an open circuit (or, as radio men say, is not "loaded").

To provide a signal voltage step-up, the interstage transformer should have the highest possible turns ratio. However, the requirements imposed by the wide audio frequency band limits the amount of the turns ratio. Some, at a sacrifice of fidelity, go as high as 5-to-1, but most stay down around 2-to-1 or 3-to-1. Compare this with power transformers, which in high voltage work can easily have a 30to-1 step-up ratio! Let's see what factors set these limits.

PRIMARY INDUCTANCE

A typical interstage transformer connection is shown in Fig. 9. Since no power is transferred, the transformer, by itself, could give us a primary to secondary voltage step-up that would be almost constant over a rather wide band of frequencies. However, there are several factors that limit the frequency response of the whole circuit.

One of these is the fact that the a.c. plate voltage (μe_g) of tube VT_1 divides between the a.c. plate resistance r_p of the tube and the reactance of the primary of the transformer. The plate resistance is almost con-

stant at audio frequencies, but the primary winding. since it is an inductance, has a reactance that increases with frequency. Therefore, the proportion of the a.c. plate voltage that appears across the primary depends on the frequency: more appears across it at high frequencies. less at low frequencies. The only way we can keep this from causing serious distortion is to make the reactance of the primary fairly high with respect to the a.c. resistance of the tube at all frequencies for which the system is designed. Then we will always have a reasonable proportion of the signal voltage appearing across the primary. There will be less at low frequencies than at high frequencies, but there will be a fair amount at all frequencies in the design range.

This means that we must have a high-inductance primary—one with



FIG. 9. Typical connections for an interstage transformer.

many turns. The higher the plate resistance of the tube, the more inductance we need in the primary to make its reactance comparable to the plate resistance. As a practical matter, since the amount of inductance we can have in the primary is limited, we have to use a tube that has a low plate resistance; otherwise, the primary inductance would have to be impossibly large. Generally speaking, only triodes have sufficiently low plate resistances; transformers are used rarely with other tubes.

► The inductance of the primary can be made high by using a large iron core or by using many turns of wire on the primary. However, the use of a large corc increases the cost, weight, and size of the transformer greatly. All these factors must be considered in radio equipment, so the necessary high inductance is usually obtained by using a large number of primary turns. There are limits here, too, when voltage gain is desired. If the primary has many turns, a high stepup turns ratio would mean thousands tiny capacity exists between the turns of wire on a transformer winding, and a somewhat larger capacity exists between layers of wire when they are wound like thread on a spool. These eapacities add up; the more turns and the more layers, the greater the capacity. This winding capacity is called *distributed capacity*, from the manner in which it is distributed throughout the transformer.

The distributed capacities can be lumped together and represented by a single capacity across each transformer winding, as shown in Fig. 10. In addition, C_P and C_S can be used to represent any stray capacities between the circuit wires, as well as in-



FIG. 10. Notice how the primary distributed capacity C_P acts as a voltage divider with r_P .

of secondary turns, requiring a great amount of wire and a large core. Thus, a transformer designed to have a good low-frequency response and a high turns ratio at the same time would have to be very large and expensive.

DISTRIBUTED CAPACITY

At times, it may seem worth the price to get a transformer with a high turns ratio. But the very fact that there are a large number of turns introduces another factor that limits the frequency range such a transformer can handle.

As you know, a capacity exists between any two conductors that are separated by an insulator. A very ternal tube capacities.

As you can see from Fig. 10, the primary capacity C_p forms a voltage divider with the plate resistance r_p . The reactance of a capacity goes down as the frequency goes up. Therefore, as higher audio frequencies are fed into the circuit, more and more of the source voltage will be dropped in the plate resistance of the tube instead of appearing across the primary of the transformer T. In fact, at very high frequencies, C_P can actually act as a short-circuit across the primary.

 \blacktriangleright Careful design can remove some of the undesirable effects of C_P . For example, this capacity and the inductance of the primary winding can be so arranged that they form a parallel resonant circuit that has a very high impedance at its resonant frequency. If this is done, practically all the a.c. voltage μe_g that is at or near the resonant frequency will appear across the primary instead of being dropped in the tube resistance. However, at frequencies above this resonant point, the circuit will cut off sharply.

But even the best design cannot eliminate distributed capacity altogether; and, unfortunately, it is always true that attempts to increase the inductance by winding on more turns will increase the distributed capacity also. In other words, we can increase the gain of a transformer only by decreasing its fidelity (its ability to transfer all frequencies equally well). Again, we must compromise. We can have a transformer that has high gain and low fidelity, or low gain and high fidelity, or medium gain and medium fidelity, but we can very seldom get both high gain and high fidelity.

SATURATION

When a transformer primary is connected into the plate circuit of a vacuum tube, we introduce the problem of saturation. The d.c. plate current of the tube must flow through this primary; if this current is too large, it will produce core saturation and reduce the primary inductance greatly. For this reason, the amount of permissible d.c. current flow is specified for each transformer, and the air gap of the core is fixed for this value.

SUMMARY

Obviously, an interstage trans-

former is carefully designed. You can appreciate why the high-fidelity types are bulky and very expensive they have to be wound in special ways to control the distributed capacity and get_the proper_inductance_ Even with special design, it is impractical to have a high turns ratio and have a high fidelity response at the same time, because there are limits to the amount of correction that is economical. Therefore, all high-fidelity interstage transformers will have a low step-up turns ratio; usually it will be less than 3-to-1

Today, high-gain tubes are available, and the increased gain from these tubes has made the voltage amplification of the interstage transformer unnecessary. This fact, plus the shortcomings of interstage transformers, have together practically eliminated this style of transformer from the average receiver. Other coupling methods offering higher fidelity are used instead. However, interstage transformers still are used where gain is important; where some special circuit action is desired; and where weight and cost are not vital factors.

▶ In your work, you will have occasion to replace these transformers as they become defective. Do not put in just any replacement; you must choose a transformer having characteristics similar to the original. Fortunately, transformers are standardized to a great extent so it is possible to get a duplicate or an acceptable substitute from the listings of supply catalogs. In these, you will find transformers listed according to the model number of the sets or other equipment in which they can be used; thus, it is easy to select the proper replacement.

Impedance-Matching A. F. Transformers

The entire object of amplification in a radio is to get enough signal power to operate the output devicewhether it be a loudspeaker, a relay, a recorder, or something else. As the power needed is relatively small (1 to 20 watts in the average radio), a single power output stage is all that is needed, provided we can amplify the signal voltage enough to operate this output stage. For this reason, voltage amplifiers are used ahead of the output stage to deliver to it the required grid voltage. (The output stage may use two tubes, as you will learn later, but only one stage of power



FIG. 11. This basic circuit is used to prove the necessity for impedance matching when the maximum power output is wanted.

amplification is common today.) ▶ In the power stage, a reasonable efficiency must be obtained so that the output actually delivered will be sufficiently large without requiring too much B-supply power. We can get the greatest transfer of power to the load when the load impedance equals the tube a.c. plate resistance, but it rarely happens that the actual load (the loudspeaker in a receiver) comes anywhere near the tube plate resistance value. However, we can use an impedance-matching transformer to make the two values appear alike so that the maximum power output can be obtained.

▶ Before we learn how this is done,

let's review the need for impedance matching by considering a few examples.

Suppose we have the circuit shown in Fig. 11, where the source resistance is R_s and the load is R_L . Let's assume that E is 12 volts and R_s is 2 ohms, then let us vary R_L .

Case 1. First, let's make $R_{\rm L}$ equal 1 ohm. Then:

 $I = \frac{E}{R_s + R_L} = \frac{12}{2+1} = \frac{12}{3} = 4 \text{ amps.}$ The voltage across R_L is then:

 $E_{\rm RL} = I \times R_{\rm L} = 4 \times 1 = 4$ volts. Now, the power developed in $R_{\rm L}$ is:

 $P_{\rm RL} = I \times E_{\rm RL} = 4 \times 4 = 16$ watts.

Thus, when $R_{\rm L}$ is 1 ohm, 16 watts will be delivered to it under the assumed conditions.

Case 2. Let us now make $R_{\rm L}$ equal 2 ohms and go through the same procedure to find the power output.

 $I = \frac{E}{R_s + R_L} = \frac{12}{2+2} = \frac{12}{4} = 3 \text{ amps.}$ $E_{RL} = I \times R_L = 3 \times 2 = 6 \text{ volts.}$ $P_{RL} = I \times E_{RL} = 3 \times 6 = 18$ watts.

This is more power than that of Case 1.

Case 3. Let us now make $R_{\rm L}$ equal 3 ohms and find the power output.

 $I = \frac{E}{R_s + R_L} = \frac{12}{2+3} = \frac{12}{5} = 2.4 \text{ amps.}$ $E_{RL} = I \times R_L = 2.4 \times 3 = 7.2 \text{ volts.}$

 $P_{\rm RL} = I \times E_{\rm RL} = 2.4 \times 7.2 =$ 17.28 watts.

We can now compare the three conditions: Case 1 has R_L less than R_s ; Case 2 has R_L equal to R_s ; and Case 3 has R_L greater than R_s . The circuit current drops as R_L is increased, while the voltage across R_L rises as R_L is increased, as we would expect. Notice the power, however. It is greatest when $R_{\rm L}$ equals $R_{\rm s}$, and is less than this maximum if $R_{\rm L}$ is made either larger or smaller. Thus, we get a rule: The maximum power output is obtained when the source and load impedances are matched_or equal.

As a further example of the need for matching, suppose we have a circuit like that shown in Fig. 12A, with a load connected directly into the plate circuit of a tube. When this circuit is drawn in the equivalent form



FIG. 12. When the load equals the 2000-ohm plate resistance, maximum power output is obtained.

shown in Fig. 12B, $R_{\rm L}$ represents the load resistance, the generator $\mu e_{\rm g}$ represents the a.c. plate signal voltage, and $r_{\rm p}$ represents the plate resistance of the tube. Let us suppose $r_{\rm p}$ equals 2000 ohms and that $R_{\rm L}$ is variable, and let's see how the power absorbed by $R_{\rm L}$ varies as its resistance is increased from, say, 500 ohms to 10,000 ohms.

The curve in Fig. 12C shows the power developed across the load under these conditions. Notice again that the maximum power is developed when the load resistance and the plate resistance of the tube are equal (or *matched*). If they are not exactly equal, considerable power still will be developed across the load—but if they are far different, particularly if the load resistance is much-less-than the plate resistance, then the power developed across the load will be reduced greatly.

► This at once introduces a problem, as the load is rarely equal to the source value in radio circuits. For example, suppose we have the following problem: A triode output tube having a plate impedance of 6400 ohms is to supply power to a 4-ohm speaker voice coil. Obviously, from what you've just learned, these impedances are so poorly matched that very little power will be developed across the speaker coil. We can, however, apply maximum power to the speaker if we couple it to the tube through an impedance-matching transformer.

To see why a transformer can match (that is, equalize) impedances, let us go back to the previously explained power transfer effect of a transformer. Let us suppose the 4-ohm voice coil has 1 aupere of current flowing through it. Its voltage will then be 4 volts and the power it absorbs will be 4 watts. The primary of the impedance-matching transformer must take this power from the source.

Turning now to the tube, we can assume that its a.c. plate voltage will be high. In fact, taking a typical example, the a.c. grid voltage e_g may well be as much as 30 volts, so if the tube μ is 6, there could be as much as 180 volts (30 × 6) in the plate circuit. Clearly, we need a step-down transformer, as the load only requires. 4 volts.

Let's assume we have a step-down transformer with a ratio of 40 to 1.

(A step-down transformer can have a far larger turns ratio than a step-up type as it requires *fewer* secondary turns—not more.) As you know, the ratio of primary to secondary voltage depends on the turns ratio, so, with a 40-to-1 transformer and a requirement of 4 volts across the secondary, the primary voltage must be at least 4×40 , or 160 volts. Under this condition, the load is absorbing 4 watts (at least) from the source. We can now find the primary current $(I_p = P \div E)$. This gives us $4 \div 160$, or .025 ampere.

Now, knowing the voltage across the primary and the current that must pass, we can find the primary impedance. From Ohm's Law, $Z = E_P \div$ $I_P = 160 \div .025 = 6400$ ohms. At the beginning, we assumed a tube impedance of 6400 ohms, so the tube and the effective primary value are matched or equal. Thus, when the connections are as shown in Fig. 13, the tube "sees" the 4-ohm loudspeaker through a 40-to-1 transformer as a load of 6400 ohms, so it delivers the maximum power to the transformer for transfer to the speaker.

► Of course, we assumed the proper transformer in our example. If we use some other turns ratio or a different load value, we will have a different primary matching value. Thus, if we use a 30-to-1 transformer with a 4-ohm voice coil, we would have a primary impedance of 3600 ohms, while a 6-ohm voice coil with the 40to-1 transformer would offer an impedance of 9600 ohms to the tube. Thus, regardless of the tube's plate impedance or the load impedance, they can be matched if a transformer of the correct turns ratio is selected.

 \blacktriangleright In your service work, you won't have to figure the turns ratio needed to match a tube to a load, as the

manufacturers list transformers for the tubes and speakers (or other loads) with which they will work properly. You can find the proper replacement transformer just by looking up the right tube and load. However, if you are interested in figures, the footnote * below gives the method of calculating the turns ratio.

Radio men often call the effective impedance of a primary when a load is connected to the secondary the "reflected impedance" of the load. Thus, in our example, the 4-ohm load caused the primary of a 40-to-1 transformer to have an effective impedance of



FIG. 13. The impedance-matching transformer causes the load to act in its primary circuit as the required impedance value.

6400 ohms. This 6400-ohm figure is called the "reflected impedance" of

* Dividing the source impedance by the load impedance gives a number that is the same as that found by multiplying the turns ratio by itself. That is, $N \times N$, or $N^3 =$ $Z_A \leftrightarrow Z_L$. From this, the turns ratio is equal to the square root of the result of dividing the source impedance by the load impedance. Expressed as a formula:

$$N = \sqrt{\frac{Z_{A}}{Z_{L}}}$$
 where

 Z_{\perp} is the source impedance (connected to the primary), Z_{\perp} is the load impedance (connected to the secondary), and N is the ratio of primary-to-secondary turns.

Using the figures of our example, we have.

$$\sqrt{\frac{6400}{4}} = \sqrt{1600} = 40$$
, so

the 40-to-1 ratio is the proper one for this particular tube and speaker.

the 4-ohm load. It is equal to the load impedance multiplied by the square of the turns ratio.

► As you will learn in another lesson, an exact match of power tube to load will not give the greatest undistorted power output. In fact, the maximum undistorted power is obtained when a triode tube feeds into a load twice its plate impedance, while with a pentode, a value near one-seventh its impedance is the best value. Transformer lists take these facts into consideration; when you select a replacement from a list, it will give maximum undistorted power for the particular tube and load for which it is intended.

FIDELITY FACTORS

The fidelity, or frequency response, of a stage containing an impedancematching transformer is affected by three factors with which the transformer itself is directly concerned. These are the design of the primary, flux leakage, and distributed capacity.

PRIMARY INDUCTANCE

High primary inductance is just as important in an impedance-matching transformer as it is in an interstage transformer. This will be easier to understand by referring to Fig. 14,



FIG. 14. The load "reflects" in the primary circuit as R_A. The primary winding must have a high reactance at the desired frequencies so that it will not reduce the effective load value.



FIG. 15. The turns not used to actually transfer power are represented here as leakage inductances.

which is the equivalent circuit for Fig. 13. In effect, the transformer "reflects" the value of $R_{\rm L}$ into the plate circuit as $R_{\rm A}$ —a value which is equal to $R_{\rm L}$ times the square of the turns ratio of the transformer.

As you will notice, R_A is effectively in parallel with L_P . When the primary reactance is very high compared to R_A , the parallel combination effectively has the value of R_A , so the tube feeds into the impedance for which the circuit is designed. We then say that the reactance of L_P has "disappeared," because it has no effect on the circuit.

However, as you know, inductive reactance goes down when frequency is decreased. Therefore, at low frequencies, it is quite possible for the reactance of L_P to be a rather low value compared with R_A . When this happens, the effective load is reduced; then the tube a.c. resistance and the load are no longer well matched, and more of the signal energy is lost in r_p .

To prevent this from happening, the primary inductance must be designed to have a reactance that is five to ten times the tube a.c. plate resistance. Then, when the secondary of the transformer is properly loaded, the primary reactance "disappears" even at low frequencies and is replaced by the reflected load R_{A} .

LEAKAGE INDUCTANCE

The fact that power is transferred in an impedance-matching transformer means that a high primary current flows (compared to that in interstage transformers), and, in turn, this means a higher flux density in the core. This condition tends to produce core saturation, which increases the leakage flux. We have already described leakage flux as the flux taking an air path that fails to link with all (or a part) of a winding. The turns that produce this leakage flux are really not a part of the transformer, for they do not assist the transformer action. We can therefore consider each transformer winding to be made up of two parts, as shown in Fig. 15. One section of each winding represents the turns actually engaged in producing flux linkage $(L_{\rm P} \text{ and } L_{\rm S})$, and the other represents the inductance producing the leakage flux $(L_{P1} \text{ and } L_{S1})$. The latter are called leakage inductances, and are considered to be in the positions shown in Fig. 15.



FIG. 16. The distributed capacity C_P can be made to resonate with the primary, which improves the high-frequency response up to the resonant frequency. However, above this point, there is a sharp drop in response.

The reactance of L_{P1} naturally reduces the amount of signal voltage that will be developed across R_A . Furthermore, since the reactance of L_{P1} increases with frequency, the voltage across R_A will decrease as the fre-

quency is increased. In the secondary circuit, the reactance of L_{s1} reduces the voltage applied to the load.

The effects of these two leakage inductances, then, are to reduce the useful signal voltage in the circuit and



FIG. 17. Both an interstage and an impedancematching transformer are shown in use here.

also to create distortion (since highfrequency voltages are reduced more than low-frequency voltages).

DISTRIBUTED CAPACITY

An impedance-matching transformer has distributed capacity, just like an interstage transformer. As shown by Fig. 16, the distributed primary capacity $C_{\rm P}$ forms a parallel resonant circuit with the primary and its leakage inductance. Similarly, the distributed secondary capacity forms a series resonant circuit with the secondary and its leakage inductance. However, in a step-down transformer with a high turns ratio (which is what most impedance - matching transformers are), secondary leakage inductance and secondary distributed capacity are so small that they can be ignored. ► Sometimes both interstage and impedance-matching transformers are used in the same apparatus. For example, in Fig. 17 transformer T_1 steps up voltage while T_2 matches the loudspeaker voice coil to the plate resistance of VT_2 .

World Radio History

Special Transformers—Identifying Types

There are one or two special types of transformers which are of interest. We will introduce them briefly here their uses will be covered in another lesson.

CENTER-TAPPED TRANSFORMERS

Fig. 18A shows the polarity that may exist, at a given instant, across the secondary of an ordinary audio transformer. This polarity will be both a positive and a negative voltage —that is, two voltages 180° out of phase with each other—from the transformer at the same instant of time.

You will see this principle used many times in radio work. It is used in power supplies to obtain full-wave rectification and in audio amplifiers to drive a push-pull stage (a type of amplifier) from a single tube stage.



FIG. 18. Many transformers have a center tap on the secondary as at B. This provides a twophase output that is necessary for the operation of certain radio stages.

reversed on alternate half cycles. As there is but one output voltage, technicians say that this is a *single phase* output.

In Fig. 18B, a center tap has been added to the secondary. The addition of the center tap does not change the voltage induced in the secondary by the primary, but does give us another point from which to measure. Thus, by measuring from this center tap to either of the original secondary terminals, we will get half the total secondary voltage. And notice-when point A is negative with respect to point C, point B will be positive with respect to point C. The next instant. the polarities will reverse, but they will remain opposite to each other at all times. Thus, if we use the centertap C as a reference point, we can get

We will say more about these transformers when we study these stages.

AUTO-TRANSFORMERS

The schematic diagram of a special type of transformer known as an auto transformer is shown in Fig. 19. (Auto means "self," the name coming from the fact that a single winding is used as a transformer.) You will notice there is a tap on the winding: primary voltage is applied between this tap and one end of the winding, and secondary voltage is taken off between the two ends of the winding. The number of turns in the entire winding divided by the number of turns connected across the input, or primary terminals gives the voltage step-up ratio of this transformer. An autotransformer also may be used as a

step-down transformer by connecting the source voltage to the ends of the winding and taking the output from the tap on the coil and one end of the winding.

While an auto transformer has certain advantages (principally low cost) to recommend it, there are many circuits in which it cannot be used because of the common connection between the source and the load.

IDENTIFYING IRON-CORE CHOKES AND TRANSFORMERS IN RADIO RECEIVERS

Power Transformers. When the receiver is an a.c. type using a power transformer, you generally can assume that the largest iron-core device is the power transformer, for it must handle all the filament power and the electrode power required by the tubes in the receiver. (However, remember that there are many receivers, principally a.c.-d.c. sets, that do not use a power transformer.)

When you must be sure of your identification, remember that a power transformer has a great many terminals or leads. You will find wires running from it to the filament terminals of each tube socket. You will find two wires running from the power transformer to the plates of the rectifier tube (assuming it is a full-wave rectifier tube). Two more wires will run directly to the line cord, one connecting through the on-off power switch of the receiver.

Power transformers have a primary winding and several secondaries—a high-voltage secondary winding, and one or more filament supply secondary windings. The low-voltage filament supply secondaries will have the *lowest* resistance because they handle high currents. The primary winding will also be low in resistance, but not as low as the filament secondaries, because it has many more turns of wire than they do.

The high-voltage secondary winding of a radio receiver power transformer can always be identified with an ohmmeter, for it will have the highest resistance of all windings on the core. This winding usually will have a center-tap terminal located between the two outer terminals of the winding; naturally the resistance between this mid-tap and an outer terminal will be about half that between the two outer terminals. (While the tap



FIG. 19. An auto transformer has a single winding which serves both as the primary and the secondary.

is located so that the number of *turns* on each section of the winding is equal, the half of the winding that is closer to the iron core will use turns of smaller diameter. This requires less wire, so this half of the winding will have a lower resistance than the outer half of the winding.)

Iron-Core Chokes. You can distinguish iron-core chokes from ironcore transformers in radio receivers by counting the terminals (or leads). Iron-core *chokes* will have two or three terminals (or leads), while *transformers* have at least four leads. (Auto transformers may have only three leads, but are seldom found in radio receivers.) Iron-core chokes used in radio receivers are of two types: those designed to handle a.f. signal currents, and those designed to filter the rectified current in the power pack. Signal current chokes are rare in the average modern receiver, so it is safe to assume any iron-core choke is a filter choke, unless you find a *direct wire* connection from this choke to the grid or plate of a signal circuit tube.

Audio Transformers. Audio transformers are usually the hardest to identify. It is safe to assume that a receiver that uses a dynamic loudspeaker always has an output transformer or impedance-matching transformer between the loudspeaker and the output stage. Generally, this transformer will be mounted on the frame of the loudspeaker, but it may be mounted on the chassis.

Since the impedance of the average loudspeaker is much lower than the a.c. plate resistance of an output tube, a step-down ratio is required in the output transformer. The primary winding therefore will have a much higher resistance than the secondary winding (which connects to the loudspeaker). Most other a.f. transformers you encounter in receivers will have voltage step-up ratios, and can be identified through their connections to the grids and the plates of amplifier tubes. The primary winding of an interstage audio transformer will have a lower resistance than the secondary winding. Transformers of this type are heavier and usually much larger than output transformers, because of the larger secondary.

► These are merely general suggestions; for further information on a particular receiver, refer to the schematic circuit diagram and service instructions for that receiver.

Looking Ahead. Since you now have mastered the important fundamental features of most of the parts used in radio equipment, and have acquired a good idea of how these parts perform in vacuum tube circuits, you are ready to study in detail the uses for these vacuum tube circuits in radio apparatus. In the next lesson you will find a wealth of interesting and highly practical information on the power supply systems used in radio.

THE N. R. I. COURSE PREPARES YOU TO BECOME A RADIOTRICIAN & TELETRICIAN (REDISTRED U.S. PATENT DIFICIO (REDISTRED U.S. PATENT DIFICIO



Lesson Questions

Be sure to number your Answer Sheet 11FR-2.

Place your Student Number on every Answer Sheet.

Send in your set of answers for this lesson immediately after you finish them, as instructed in the Study Schedule. This will give you the greatest possible benefit from our speedy personal grading service.

- 1. What is the opposition effect in a magnetic circuit called?
- 2. What is the *practical unit* of magnetomotive force for coils?
- 3. What condition is said to exist in an iron core when increases in magnetomotive force produce little or no increase in magnetic flux?
- 4. How can eddy current losses be reduced considerably in low-frequency ironcore devices?
- 5. Does primary leakage flux induce any voltage in the secondary winding of a transformer?
- 6. Suppose a circuit uses a step-down transformer. Will the primary CUR-RENT be: 1, the same as; 2, larger than; 3, smaller than the secondary CURRENT?
- 7. Would you expect a high-fidelity interstage transformer to have: 1, a high; or 2, a low step-up turns ratio?
- 8. Is the maximum power transferred when the load impedance is: 1, equal to; 2, greater than; or 3, less than the source impedance?
- 9. When checking the various windings of a radio receiver **power** transformer with an ohmmeter, which winding will have the highest resistance?
- 10. Although the number of turns on the two halves of a high-voltage winding of a power transformer are equal, are the resistances of the two halves exactly the same?



HOW TO CONCENTRATE

The secret of rapid progress with any course of study lies in being able to concentrate. If your mind wanders from study while you are alone in a quiet room, try moving to a noisy room, or try tuning in an all-musical program on the radio. Stay away from loud conversation, though.

If noise definitely bothers you, however, do your studying in a quiet location. If you feel too sleepy to concentrate, the room may be too hot. Open the windows, and put on a coat if necessary, for it is easier to study in a cool room. Sponge your face, neck and eyes with cold water.

If you have difficulty in understanding a subject, write down an explanation of it in your own words, or try outlining the subject and studying your outline. Underlining of important words as you study is another aid to learning.

It will pay, too, to experiment until you find the conditions that are most favorable for concentration when you study. Also, try different studying techniques, until you find the one that gives you complete mastery of a subject in the shortest possible time. The faster you master each lesson, the sooner you'll complete the Course and be ready for a good job in radio.

J. E. SMITH





STUDY SCHEDULE NO. 12

For each study step, read the assigned pages first at your usual speed. Reread slowly one or more times. Finish with one quick reading to fix the important facts firmly in your mind, then answer the Lesson Questions for that step. Study each other step in this same way.

- □ 4. Taking out the Ripple......Pages 12-20 The pulsating d. c. from the rectifier is a combination of d. c. and a. c., so the a. c. must be removed for proper radio performance. The standard filter is a coil-condenser combination, arranged so most of the a. c. is dropped in the coil. The choke input and the condenser input types differ in their operation, d. c. voltage output and effects on the rectifier. This section should be studied and restudied. Refer back to it later on, as many very important facts are presented here. Answer Lesson Questions 5, 6 and 7.
- □ 6. Facts About Rectifier Tubes......Pages 25-28 There are two general types of rectifiers—vacuum types and the mercury vapor types. The difference in regulation and current handling ability should be carefully noticed, so the wrong tube won't ever be used. Answer Lesson Question 10.
- 7. Mail your Answers for this Lesson to N. R. I. for Grading.
- □ 8. Start Studying the Next Lesson.

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World Radio History

HOW OPERATING VOLTAGES ARE OBTAINED FROM AN A. C. POWER LINE

Power Supply Requirements

WHEN you become an expert at tracing circuits, whether you are working on a receiver, a transmitter, or some sort of electronic control, you will automatically divide the device into two parts:

1. The signal circuits.

2. The power supply circuits.

These two circuits can always be considered separately. In fact, special filters are almost always used to keep them from reacting on one another.

One very important fact you must remember about any modern radio receiver is that it uses the tiny amount of electric power picked up by the antenna to excite a tube or tubes. These tubes in turn control a relatively large amount of power furnished by the batteries or other power source. The circuits which control the power are called the signal circuits—the circuits which furnish the power are called the power supply circuits.

Both kinds of circuits use resistors, condensers, coils, transformers and tubes. You've already studied the basic actions of these important radio parts. Now you're ready to learn all the facts about their many uses in radio circuits. It's this knowledge, which you'll gain from this and succeeding lessons of your N.R.I. Course, that will make you an expert technician—able to operate and repair all types of radio equipment.

A great many of your later lessons

will be devoted to the signal circuits of modern radios. In this lesson, you're going to learn all about the equally important power supply circuits, without which no radio could operate. You should study this lesson very carefully, for the information it gives you will be useful all through your radio career.

WHAT THE POWER SUPPLY MUST DO

A radio tube works best when the voltages applied to it have proper polarities and values. The plate, control grid, screen grid and suppressor grid electrodes all require different operating voltages with respect to the cathode of the tube. As you've already learned, these voltages must make the screen and plate positive with respect to the cathode, and must make the control grid and suppressor grid negative with respect to the cathode. (In some cases the suppressor grid is connected directly to the cathode, and then there is no difference in potential between them.) The filament of the tube must be heated so that electrons will be emitted. You must apply just the right voltage to the filament to heat it properly.

▶ Thus, one important requirement we must make of the power supply is that it furnish the proper operating voltages for the various tube circuits. Another is that the power supply must not interfere with the operation of any stage by introducing undesirable

1

voltages. Only the signal voltage applied at the input of the tube should control the output of the stage. We do not want supply voltages to have a ripple component which might act just like a signal and make the loud-speaker produce noise or hum. For this reason, the power supply must be a pure d.c. power source with a minimum of variation or ripple.

Finally, we do not want the power supply to transmit signals from one stage to another. You can easily see that signals straying through the power supply would be as bad as having variations in the power supply.

Let's repeat these three important things required of the power supply, in order to fix them in your mind:

1. It must provide the correct amount of voltage to each tube electrode. Each voltage must have the correct polarity. (As you will soon learn, these voltages may have to be values between 1 or 2 volts and several hundred volts.)

2. It must furnish as near a pure d.c. voltage as possible to those electrodes where variation would produce distortion, interference or unwanted signals.

S. It must not transmit signals from one stage to another. This means that the power supply circuits and the signal circuits can be, and are, electrically separated from each other—which, as you've learned, is the reason we can study them individually.
▶ Of course, these requirements must

be met by any power supply which may be used. Now, let's see what kinds of power supplies have been developed.

BATTERY SUPPLIES

Batteries meet all the requirements of a power supply—cells can be added to get the required voltage; a pure d.c. is delivered; and their low internal resistance causes little signal voltage drop, thus minimizing unwanted signal transfer between stages. ▶ In fact, batteries are such perfect sources of d.c. power that engineers frequently draw battery symbols in their diagrams to show that a d.c. supply is wanted.

► A simple battery supply circuit is shown in Fig. 1A. The signal circuit parts which you would ordinarily expect to find in the control grid and plate supply leads have been omitted to simplify this drawing. These parts



FIG. 1. Power supply connections for a triode and a pentode are shown in A and B respectively. Bypass condensers C_1 and C_2 are used to keep stray signals out of the power supply.

-which might be resistors, coils or transformer windings-would be connected between the points which are connected by the dotted lines.

In Fig. 1A, we have an A battery supplying power to operate the filament, a B battery to operate the plate circuit and a C battery to supply the necessary grid bias. These batteries are usually made up from dry cells, although a storage battery or an aircell battery may be used as an A supply.

► Although batteries have these very desirable characteristics, they are ex-

pensive, bulky, and must be changed or recharged at frequent intervals. Therefore, as soon as radio became popular, many experts searched for ways to eliminate batteries by converting a.c. from the power line wall outlet into continuous or d.c. power. At first, they tried to get pure d.c. for all tube elements, including fila-However, the high-current, ments. low-voltage supply needed for filaments demanded special, expensive equipment-so an a.c. type tube, whose filament could use a.c. directly, was developed.

High d.c. voltage is easily produced from the a.c. power line, particularly if the current demand is not too high. Once we have a high d.c. voltage, it is a simple matter to divide it so that the grid and plate elements receive the proper amount of voltage.
▶ Let us now see how the power supplies work which use power transformers to operate from standard a.c. power lines. First, we will take up filament voltage supplies, then study the rectifiers, filters and voltage dividers used to obtain power for the other tube elements.

Filament Power

A d.c. voltage applied to a tube filament will neither cause changes in the plate current wave form nor introduce hum. However, when a.c. is applied, we must take special precautions to make sure the filament heat does not vary and to prevent the a.c. from affecting the grid bias and plate voltage. Let's see what these precautions are for a tube which uses a filament as the cathode or electron emitter.

FILAMENT TYPE TUBES

Getting Constant Heat. When a.c. is fed to the filament, the current rises to a peak twice for each cycle, once on the positive and once on the negative alternation. The heating effect is the same on either alternation. Thus, the filament will be subjected to a maximum heating current 120 times a second if standard 60 c.p.s. (cycles per second) current is used. If the filament is very thin, as it is in most battery-operated tubes, it will become hot, then cool off, 120 times a second. Since, as you recall, the number of electrons emitted from a filament depends on how hot the filament gets, this varying filament heat will produce a varying electron flow from the filament to the plate. This pulsating plate current, varying 120 times a second, will produce a loud hum.

We can prevent the filament heat from varying by using a filament so thick that, once it is warmed up, it tends to stay at a constant heat. This is exactly how a.c. filament tubes are made.

▶ You can now see why battery type tubes are not satisfactory when their filaments are powered from an a.c. source—and also why battery receivers using tubes with thin, quicklyheated filaments go into operation faster than a.c. receivers using heavy filaments which take a longer time to heat.

► A heavy, hcat-holding filament solves only one of the problems involved in a.c. filament operation. Still other precautions must be taken to prevent an a.c. filament supply from producing a.c. ripple in the plate current. Getting Constant Emission. Now let us see how an a.c. filament voltage can affect the grid voltage. Look at Fig. 1A. Here we can measure the grid voltage by touching the negative probe of a voltmeter to the grid G and the positive probe to -A. Since +C connects to -A, the voltmeter is connected across the C battery and measures the C battery voltage.

Suppose, however, that we connect the positive voltmeter probe to +A. Will we measure the same grid-filament voltage as before? No-because now the A battery voltage will add to the C battery voltage, and we will actually measure the C battery voltage plus the A battery voltage.

Since the grid is more negative with respect to the right-hand side of the filament than to the left-hand side, the number of electrons leaving the right-hand side of the filament is considerably less than the number leaving the left-hand side.* We don't care about this when d.c. is used on the filament, because the total emission remains constant, and the plate current is not varied by the filament voltage.

When a.c. is used to heat a filament, we cannot connect +C to either side of the filament. Fig 2, where such a connection is made, shows you what would happen. Here the plate has been omitted for simplicity, and the filament is shown as a straight resistance wire because it has resistance. This representation will help you to picture the voltage drop that exists across the filament when filament current flows through it. The C battery furnishes 3 volts d.c., while 6 volts a.c. are used to heat the filament. At the particular instant illustrated, the a.c. voltage gives the filament the polarity shown.

Starting with Fig. 2A, the voltage between the grid and point a on the filament is -3 volts. As we move along the filament, the grid-filament voltage becomes progressively greater as the voltage drop along the filament resistance is added to the bias voltage. Between the grid and b it is $-4\frac{1}{2}$ volts; at c (the center of the filament) it is -6 volts, and at d, where all the filament voltage drop is added to the bias voltage, it is -9 volts. Naturally, most of the electrons reaching the plate come from the section between a and c, because the grid is less negative with respect to this part of the filament.

If the supply current is 60-cycle a.c., the filament voltage will reverse polarity so that in 1/120 of a second we will have the condition illustrated in Fig. 2B. The voltages between the grid and various points on the filament at this instant are indicated in the diagram. As you see, the filament voltage drop is now subtracted from the 3-volt grid bias voltage. This creates an over-all decrease in gridfilament voltage-and so considerably increases the total number of electrons reaching the plate, as compared with the number which reached it during the previous half cycle.

In another 1/120 of a second, the conditions are again as shown in Fig. 2A. As a result of these changes in grid-filament voltage, the plate cur-

^{*} When d.c. filament voltage is employed, this results in extra wear on the side of the filament emitting most of the electrons. This effect is unimportant in low-power receiving tubes, but in transmitting tubes where the plate current may be very high. the connections to the filament leads of large tubes are disconnected and reversed about once a week in order that both sides of the filament may receive equal wear. When the filaments of low-power tubes are fed with d.c., as in Fig. 1A, the +C and -Bconnections of the electrode voltages may be connected either to -A or +A filament terminals; the -A terminal connection has become standard for most circuits.

rent alternately increases and decreases. This causes hum.

It makes no difference to the plate from which part of the filament electrons come. The important thing is that the *total number* remain constant (except, of course, when a signal voltage is applied between the control grid and filament).

We can get constant electron emission by connecting +C to the center of the filament, as shown in Fig. 2C. The voltages between the grid and



FIG. 2. When a.c. is used to heat the filament, a center tap is necessary to prevent hum.

points a, b and c on the filament. when the supply voltage has the indicated polarity are shown directly above the grid symbol. For simplicity, suppose that from point a we get 2 electrons, from point b we get 8 electrons, and from point c we get 14 electrons. (Actually, of course, the number of electrons is far greater.) This makes a total of 24 electrons reaching the plate.

When the filament polarity reverses 1/120 of a second later, the grid-tofilament voltages at a, b, and c will be those shown in Fig. 2D. Now from a the plate gets 14 electrons, from b it gets 8 electrons, and from

c, 2 electrons. In all, the plate receives 14+8+2 or 24 electrons, the same number it did in 2C. Thus, a center-tapped filament allows us to maintain a constant flow of electrons to the plate throughout the a.c. filament cycles, even though the amount from any one spot on the filament is varying. (The signal voltage will, of course, still vary the plate current, because it changes the voltage between the grid and all points on the filament at one time.) ▶ The plate - to - filament voltage would vary just like the grid-to-filament voltage if we connected B— to one end of the filament. We can prevent this variation (which would of course cause a varying plate current) in exactly the same way-by also connecting B— to the center tap of the filament, as shown in Fig. 2D.

Getting a Center Tap. You see that a grid and plate connection to the center of the filament will eliminate a.c. variations in the plate current. While it is perfectly possible to build a tube with a filament center tap, this is not necessary. We need only connect +C and -B to a centertapped resistor across the filament (see Fig. 3A).

Let's see why. In Fig. 3B, we have a tube filament connected between points A and B. As you've learned, each point along the filament will have a different voltage with respect. to one end of the filament. For instance, with 6 volts applied and measuring from point A, there might be 1 volt at c, 4 volts at d and 5 volts at e.

If we connect a resistor in parallel with the tube filament between F and G, the same total voltage will be across this resistor—and there will be differences in voltage between point Fand other points on the resistor. There will be a point on the resistor where


FIG. 3. Practical ways of getting the center tap.

this difference is 1 volt. This is marked c_1 to correspond to the same point on the tube filament. Similarly, a 4-volt difference will be found at point d_1 , and a 5-volt difference at e_1 . Points F and A are at the same potential, since they're connected together by a wire.

For every point on the tube filament, there will be a corresponding point on the resistor which will have the same potential with respect to terminal A. Thus, center tap P on the resistor has the same potential with respect to terminal A as has point X, the center of the filament. Since P and X are both in the same circuit and are both at the same potential, we get precisely the same electrical effects by connecting an outside circuit to point P as we would by connecting it to point X.

Frequently, a potentiometer (variable resistor) is used across a tube filament instead of the fixed-tap resistor, as shown in Fig. 3.4. Then, if the electronic emission from the filament is not the same over its entire length because of imperfect manufacture, we can find a point on the resistor which corresponds to the emission center of the filament by moving point P to the right or left. Hum disappears or is reduced to a minimum when the correct adjustment is

made. Usually tubes are manufactured with sufficient care so that a movable tap on the resistor is not required.

▶ A center-tapped resistor is not always used, since it is less expensive to tap the electrical center of the filament winding on the power transformer instead (Fig. 3C). Here point P is the +C and -B terminal for this tube stage.

HEATER TYPE TUBES

Variations in voltage between the grid and the electron emitter may also be eliminated by using a cathode or electron emitter that is not connected to the heating source. A heater type



FIG. 4. The elements of a heater type tube.

tube has its filament surrounded by a sleeve coated with electron-emitting chemicals. This sleeve is the cathode, and emits electrons; the filament (whether fed with d.c. or a.c.) only furnishes heat to the cathode. Emission is kept constant by making the filament and cathode heavy enough to hold an even temperature. The connections for such a tube are shown in Fig. 4. Since no a.c. is dropped in or applied to the indirectly heated cathode, there is no need for centertapped resistors across the filament supply of these tubes. (Sometimes there is a center tap on the filament transformer. This lead is generally grounded to the chassis to reduce leakage and stray fields and thus further reduce the possibility of hum.)

Heater type tubes have replaced filament types almost altogether in radio receivers using a.c. power supplies. However, filament type tubes are still widely used in transmitters, in power packs, and in some output stages where a slight amount of hum does not matter.

FILAMENT CONNECTIONS

The tubes commonly used in a.c. operated radio receivers need filament voltages of 1.5, 2.5, 5, or 6.3 volts. These voltages are furnished by low-voltage windings on the power transformer.

Ordinarily, tubes needing the same filament voltage have their filaments wired in parallel and connected to the same filament winding on the power transformer, as shown in Fig. 5A. When circuit requirements make it necessary to separate filaments of tubes needing the same filament voltage, an extra winding will be provided on the power transformer secondary for the filaments which are to be operated as a separate circuit (see Fig. 5B).

▶ If tubes with different filament voltage ratings are used in the same piece of radio equipment, separate filament secondary windings must be provided for each group, as shown in



FIG. 5. Filaments must be connected according to their voltage and current needs.

Fig. 5B. A 2.5-volt tube, for example, cannot be operated from a 6.3-volt source, for the filament of the 2.5-volt tube will draw too much current and . burn out almost at once. That's something to remember in your service work—always be careful to operate a tube on the correct filament voltage.

Circuits 5A and 5B may be used with either filament or heater type tubes. With filament tubes, the center tap P must be used as the C+ and B- connection.

Obtaining D. C. from A. C.

You have learned that the filament can be supplied with a.c. directly if we use a cathode or indirectly heated type of tube, or use a tube with a heavy filament and make the proper connections. But we must use d.c. for the plate and grid supply.

To operate the average radio receiver and amplifier properly, 110volt, 60 c.p.s. power must be converted to about 400 volts d.c. without an appreciable ripple.* Theoretically, you cannot take out all the ripple, but practically, you can make it so small that it produces no noticeable hum in the loudspeaker or other output device.

Fig. 6 shows how the power pack works. The 110 volts from the power line are first raised to a high *a.c.* voltage by a voltage step-up transformer. Next, the a.c. is passed



FIG. 6. We start with a.c. and get d.c.

through a device which we call a rectifier, which allows current to flow only in one direction. A tube is generally used. A rectifier tube produces pulsating direct current, varying from a peak (the largest) value to zero. To remove the variation, the electric power is passed through an electrical filter consisting of condensers and choke coils, which smooths out the variation or ripple, leaving pure d.c. Finally, the high d.c. supply voltage is divided as required. The changes from a.c. to divided d.c. are portrayed in Fig. 6. At A, B, C, D and E, the wave shape of the output voltage of each section is shown.

You know already that a transformer with more turns on its secondary than on its primary will step up a voltage according to its turns ratio. The power transformer in the average receiver steps up the power line voltage to as much as 350 or 400 volts—ample voltage for the tubes

* By ripple, we mean a small a.c. voltage superimposed on (combined with) the d.c. used in radio receivers, amplifiers and control circuits. Of course, this a.c. voltage cannot be applied to the receiving tube electrodes, since the electrodes would be alternately positive and negative. We must rectify this a.c. supply into a d.c. voltage.

HALF-WAVE RECTIFIERS

Figs. 7A and 7B show the basic operation of a half-wave rectifier. In these circuits, the single resistor R_{L} represents all the tubes and circuits of a radio. We can use this representation because each tube draws current, and thus acts as a load on the power supply. The combined resistance of these various loads is equal to the supply voltage divided by the total current drawn. Our resistor $R_{\rm L}$ has the same ohmic value as the combined resistance of the tubes, and so has exactly the same effect as the tubes on the power supply. This resistor is called the load resistance.

Alternating voltage comes from transformer T. We want the voltage supplied to the load $R_{\rm L}$ always to have the polarity shown in Fig. 7A in other words, we want to cut out the half cycles of the supply voltage which would reverse the polarity of the voltage across $R_{\rm L}$.

We can cut out these unwanted half cycles of supply voltage by using



FIG. 7. A rectifier tube acts like an electronic switch.

some sort of switch that will close each time transformer terminal a is positive, and open each time this terminal is negative, with respect to terminal b. This can be a mechanical switch^{*} (SW in Fig. 7A), but most radios use an electronic switch.

Fig. 7B shows a diode tube VT which acts as an electronic switch. When terminal a of the transformer becomes positive, the plate of the tube is made positive, and electrons flow from the cathode to the plate. (This corresponds to switch SW in Fig. 7A being closed.) The flow ceases when the voltage at terminal a drops to zero. During the next alternation the terminal polarity reverses, making the plate negative with respect to the cathode. Now no electrons flow to the plate, and our electronic switch is open.

If the voltage across terminals a and b changes at the rate of 60 cycles per second, the voltage across $R_{\rm L}$ as the result of the rectifying action of tube VT will be a series of half-wave pulses (60 every second), as shown in Fig. 8A.

This voltage across $R_{\rm L}$ is neither pure d.c. nor pure a.c., but rather a combination of both. (We started with pure a.c. and now have the desired d.c. mixed with some a.c.; this is called a pulsating d.c.) Analysis of this voltage by a special laboratory instrument called a harmonic analyzer will show us that it consists of the d.c. component (or part) shown in Fig. 8B, the 60-cycle ripple (or a.c. voltage) shown in C, and the ripples of higher frequency and smaller amplitude shown in D and E.

Now, we want only the pure d.c. component of this voltage shown in

Fig. 8B, because the a.c. components shown in C, D and E would cause hum if we applied them to our receiver. As you will learn later in this lesson, these a.c. components can be removed by sending the rectified voltage pulses shown in Fig. 8A through an a.c. filter. This removes everything but the pure d.c., which is then applied to load resistor $R_{\rm L}$.

You can see that the d.c. component in Fig. 8B is not as great in magnitude as the peaks of the pulsating d.c. in Fig. 8A. This means that not all of the rectified voltage is available as



FIG. 8. The output of a half-wave rectifier.

d.c. for the load. Let's see how we can get more d.c. output.

FULL-WAVE RECTIFIERS

When we use the half-wave rectifier shown in Fig. 7, only one-half of the *supply* wave is rectified. (This does not mean that we are wasting power, for, when the tube is not passing current, none is taken from the source.) If we put both halves of the supply wave to work by using a full-wave rectifier, we can get a higher rectified voltage. In other words, we get a larger d.c. component from full-wave.

Further, in the half-wave rectifier, the lowest ripple frequency is equal to

^{*} Later, when auto radios are studied, you will see how an electromechanical switch called a vibrator is sometimes used to close and open power supply circuits automatically.

the frequency of the supply line, while the lowest ripple frequency of a fullwave rectifier is twice the frequency of the line (120 cycles for a 60-cycle line). This higher ripple frequency may be smoothed out with smaller, less expensive parts.

Full-wave rectification uses both halves of the a.c. cycle, as the name implies. By making a change in the transformer and by using a dual tube rectifier, we can make current from both halves of the cycle flow in only one direction through the load in our former half-wave rectifier circuit.



FIG. 9. A full-wave rectifier is a double switch.

The transformer change consists of providing the secondary with a center tap which is used as the reference terminal. When one end of the winding is positive with respect to this center tap, the other end will be negative, and vice versa.

Going back to our idea of a rectifier tube as an electronic switch, our fullwave rectifier circuit is really that shown in Fig. 9A. Here we have a switch arm c which throws to whichever transformer terminal (a or b) is positive with respect to center tap d. When a is positive with respect to center tap d, the switch closes a-c so the voltage between a and d is applied to the load.

When the a.c. cycle reverses, mak-

ing a negative and b positive, the switch changes connection and we apply voltage b-d to the load. Now let's substitute two tubes for the switch, as shown in Fig. 9B, and see how the circuit works.

When point a is positive with respect to d, electrons flow from the center tap d to load terminal y, to x, to the cathode of tube VT_1 , across to the plate and to point a. The electrons cannot go from d to b because they cannot pass through VT_2 when point b is negative with respect to d. Therefore the path taken through VT_1 is from $d \rightarrow y \rightarrow x \rightarrow cathode \rightarrow plate \rightarrow a$. On the next half cycle, when b is positive with respect to d, the path is through tube VT_2 —that is $d \rightarrow y \rightarrow x \rightarrow$ $cathode \rightarrow plate \rightarrow b$.

You see at once that the tubes alternate in passing electrons through the load $R_{\rm L}$. Also, the actual voltage rectified is half the voltage across *a* and *b*. In other words, the a.c. voltage across the entire transformer secondary in Fig. 9B must be twice as



FIG. 10. The output of a full-wave rectifier.

much as in Fig. 7B to realize about the same peak rectified voltage, because only half of the total secondary is being used at a time.

The pulsating output current through the load has the wave form shown in Fig. 10A. Compare this with Fig. 8A and you will see that there are twice as many pulses, with no gaps between pulses. This makes the d.c. average of this wave about twice as much as that of a half-rectified wave (compare Fig. 10B with 8B). Of course, we must get rid of the ripple frequencies shown in Figs. 10C and 10D before we can use this d.c. output for the load.

Full-wave rectifiers are so practical that they are almost universally used in radio receivers. Instead of two separate tubes as shown in Fig. 9B, a twin diode tube is generally used, as shown in Fig. 11. This tube has two separate diode rectifiers in a single glass or metal envelope. A rectifier tube of the cathode type is shown in Fig. 11A, while a filament type is shown in Fig. 11B. In filament types, the positive load terminal must connect to the filament. The connection may be made to either filament lead.*

(Perhaps you wonder why we connect directly to the filament instead of to a center tap on the filament winding. While it is true that the direct filament connection creates a certain amount of a.c. in the rectifier output, this is removed along with the ripple frequencies which are present in any rectifier output. The direct filament connection therefore does no harm.)

The two diode tube sections (or separate tubes, if they are used) should be "balanced"—that is, have the same emission characteristics. Furthermore, each diode should be supplied with the same amount of a.c. voltage to rectify. Both conditions make the



FIG. 11. Twin diode tubes are usually used for full-wave rectification.

filtering job easier. Should one tube pass much more current than the other, the output wave form will become more like that of the half-wave rectifier in Fig. 8A than that in Fig. 10A. In other words, the frequency of the main ripple will be cut in half; since the filter is not designed to suppress this lower frequency properly hum will be produced.

► The transformer is designed to supply approximately equal voltages to the diodes. Very little trouble can occur here. Usually an unbalance is caused by the emission of the two tube sections changing to unequal values. When you are servicing a power supply, check both sections of the rectifier tube in a tube tester and then compare the readings. If the tube sections are greatly unbalanced, you should install a new tube.

^{*} The filament voltage serves no other purpose than to make the filament hot enough to emit electrons and does not affect the B supply voltage in any way. Neither does the B supply voltage have any effect on the filament voltage. The two circuits are quite independent and are simply joined together at one point. We want to impress on you the fact that there is no interaction between the circuits, for students (forgetting that a circuit must be complete before current can flow) often ask why the high d.c. voltage does not burn out the filament. As you can see, the d.c. is not applied across the two filament leads and hence cannot cause an excessive current to flow through the filament.

Taking Out The Ripple

As you have already seen, the output of a rectifier is made up of d.c. and a number of a.c. components. We want only the d.c. to pass through the load, so we must "filter out" the a.c. in some manner. A combination of an iron-core choke and a condenser is most widely used for this purpose. Let's see how they work.

THE L-C FILTER

Fig. 12A shows the power supply circuit we have studied so far. For simplicity, S is used to represent the transformer-rectifier section, supplying both a.c. and d.c. to load $R_{\rm L}$. If we introduce a choke coil L in the circuit as shown in Fig. 12B, the d.c. and all the a.c. components must flow from the rectifier circuit through L and $R_{\rm L}$. Both a.c. and d.c. voltage drops will exist across choke coil L.

Now, you learned in an earlier lesson that a choke coil may have low d.c. resistance and high a.c. reactance. Because of its low ohmic resistance, negligible d.c. voltage will be lost in L, and most of the d.c. supplied from the rectifier will be applied to the load $R_{\rm L}$. But if the reactance of choke L is made large compared to the resistance of $R_{\rm L}$, most of the *a.c.* voltage will be dropped in the choke. If we select an inductance which will drop a large amount of the fundamental a.c. frequency, even greater voltage drops will exist for the higher frequency components-because the reactance of the choke will increase with frequency.

We could make the inductance of the choke coil so large that only a negligible amount of ripple voltage would appear across the load, but such a coil would be much too expensive and bulky. Instead, we can use an easily obtained coil and introduce a condenser into the circuit, as shown in Fig. 12C. The condenser C is chosen so that its reactance is low compared to the resistance of $R_{\rm L}$, so it acts as an a.c. short circuit across the load. As the a.e. circuit in B includes the reactance of L and resistance of $R_{\rm L}$, the low-reactance condenser in C also reduces the total a.c. circuit impedance, causing a greater a.c. flow through the choke and thus increasing the ripple voltage drop across the choke.

As far as a.c. is concerned, you can think of this circuit as containing a low-reactance condenser in series with the high-reactance choke. Since the same current flows through both, there will be a large voltage drop across the high-reactance choke, and just a small voltage across the condenser. As the load R_L is in parallel with the condenser, it also has only a small a.c. voltage across it. The filtering is just as good with this arrangement as if the choke inductance had been increased.

Another Explanation. There is another way of looking at a choke coil-condenser section such as this. The action of the coil is very simple. As you know, it always attempts to prevent a change of current—by stor-



FIG. 12. The choke input filter is formed by using a choke and condenser.

12 Id Radio History ing energy in the magnetic field when the current is increasing, and releasing energy when the current decreases. It tries to hold the current constant, so the effect of the coil by itself is to "smooth out" the current variations.

The condenser charges as the current pulse from the rectifier increases. When the pulse passes its peak and starts to decrease, the condenser discharges into the circuit. This means the condenser helps make up for the drop in current from the rectifier. Thus, like the coil, the condenser smooths out current variations; the combination of the coil and condenser naturally does this smoothing out better than either of them alone. The effect is that of cutting off the current peaks and filling in the valleys, so only an average d.c. current remains. As these L-C filters forcibly remove the a.c. component, they are known as "brute force" filters.

Summary. To sum up, the *d.c.* component of the rectifier output is divided between the *choke d.c.* resistance and the *load resistance*. Since the choke coil has a low d.c. resistance, most of the d.c. voltage appears across the load where it is wanted.

The a.c. components are divided between the high choke reactance and the low condenser reactance, so practically all the a.c. is dropped in the choke coil, leaving little a.c. across the condenser. Since the condenser and load are in parallel, this means that there will be little a.c. ripple across the load to cause hum.

▶ There is another important point we want you to see clearly. In Fig. 12C we have a series a.c. circuit consisting of the source S, L and C. As in any series circuit, the current will be the same at any point in the circuit. Therefore, if we move L to the position shown in Fig. 12D, the same current will flow through it as in Fig.

12C. If the same current flows through L, the same amount of voltage will be dropped across it and the change in position of the choke does not affect the a.c. voltage division. Hence, it does not matter whether the choke is placed in the negative side or in the positive side of the filter circuit; the same amount of filtering is obtained in either case.

MULTI-SECTION FILTERS

As you've just learned, a filter is a voltage divider. As far as the ripple is concerned, the divider consists of inductive reactance X_L and capacitive reactance X_c . The voltage divides in almost exact proportion to the reactance values, so the ripple reduction factor (ripple input divided by ripple output) equals $X_L - X_c$.

Now suppose that we have one filter which reduces the ripple 100 times, or from 100 volts to 1 volt. This means that the reactance of X_L is 100 times as great as that of X_c . Now suppose we feed this remaining 1-volt ripple into a second filter using the same values of L and C as the first. Again the filter will reduce the ripple fed to it 100 times—from 1 volt to .01 volt. The total reduction is now from 100 volts to .01 volt, which is $100 \div .01$ or 10,000 times.

Clearly, the reduction obtained from two filter sections is their reduction factor product $(100 \times 100 \text{ is } 10,000)$. To get a reduction of 10,000 times in a single filter, the choke reactance would have to be 10,000 times the condenser reactance and the expense would be prohibitive. But we can get such a reduction easily, using inexpensive parts of reasonable size, just by using two filter sections as shown in Fig. 13. Filters hooked up this way are called "cascade" filters. Cascade filters are almost always

used in transmitter power packs, where

fewer sections would mean very expensive parts. Sometimes as many as three sections in cascade are used in transmitters, high-fidelity amplifiers and special equipment. In home receivers, a single-section filter, using modern high-capacity electrolytic condensers and a good choke coil, is satisfactory in most cases.

TUNED FILTERS

You've learned that the larger the reactance L is, compared to C, for the



FIG. 13. A cascade filter.

ripple frequency we wish to suppress, the greater will be the ripple reduction. Now, a parallel resonant circuit at resonance has a greater resistance than the reactance of either the inductance or capacity used in it. We can combine these facts to produce a more efficient filter by using a circuit like that shown in Fig. 14A. Here L has been replaced by a parallel resonant circuit L_1 - C_1 through which the d.c. and all a.c. components must pass.

When L_1 and C_1 are tuned to a definite frequency, this circuit will offer a very large impedance to current of that frequency, much larger than L_1 alone. As a result, when the a.c. ripple divides between the resonant circuit and C_2 , most of the a.c. is dropped across the tuned circuit because of its high impedance at the ripple frequency.

For d.c., the resistance of the resonant circuit is simply the resistance of the wire with which L_1 is wound. This resistance is low, so most of the d.c. appears as a charge across C_2 , from which it is passed on through the next filter section to the load.

A fairly large value of L_1 should be chosen so the capacity of C_1 will be small. At frequencies above resonance, the condenser reactance is lower than that of the choke, and the higher ripple frequencies are by-passed around the choke. By keeping the capacity of C_1 small, this effect is reduced, but it is desirable to have another filter section L_2 - C_3 to be sure to eliminate the higher ripple frequencies.

Now, a series resonant circuit has a low impedance at the resonant frequency. In a filter, the shunting condenser should have low reactance. This suggests that, instead of a parallel resonant circuit in place of L, we might use a series resonant L-C circuit in place of C. Such a circuit is shown in Fig. 14B.

When L_2 and C_1 are in resonance with the lowest ripple frequency, they offer very little opposition to that component. Therefore, when the a.c. ripple divides between L_1 and the resonant circuit, most of the ripple is dropped across L_1 . This circuit is effective at the frequency for which it is designed (the lowest ripple frequency), but at higher frequencies. resonance does not occur and the circuit acts like an inductance. Thus, all component ripple voltages higher in frequency divide between L_1 and L_2 , creating a high-frequency ripple across L_2 . It is for this reason that a tuned filter must always be followed by another untuned section (L_3-C_2) in Fig. 14B).

For 60-cycle, half-wave rectifiers, the circuit L_1 - C_1 in Fig 14A or L_2 - C_1 in Fig. 14B would be made to resonate to 60 c.p.s.; for a full-wave rectifier, these circuits should resonate to 120 c.p.s. As you've learned, these are the lowest ripple frequencies found in these rectifiers.

▶ You may find a radio receiver or amplifier, using a tuned filter, which has a hum that seems to defy correction. This is often caused by a change in the inductance value of the ironcore choke (shift in the laminations from rough handling), or perhaps the filter circuit is not quite tuned to the proper frequency. To remedy this, try varying C_1 . Small condensers added in parallel will gradually increase the capacity. Should the condenser already be too large, use a smaller one in place of it and add others in parallel until minimum hum is obtained.

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If condensers C_2 and C_3 in a power supply like that shown in Fig. 14A are small paper condensers, quite often the best way to make repairs is to eliminate the tuned filter altogether by removing its tuning condenser C_1 . Then you must substitute high-capacity electrolytic condensers for the small paper filter condensers. For the circuit of Fig. 14B, remove L_2 and replace C_1 and C_2 with higher capacity electrolytics. The results with the new filters will, in many cases, be better than those with the original tuned filters in these cases.



FIG. 14. Examples of tuned filters.

▶ In each of the filters we have so far studied, the output of the rectifier has been fed into a choke. This is known as a "choke input type" filter. Voltage division has been used to cause the d.c. component of the rectifier output to appear across the load, and the energy in the a.c. components has been deliberately wasted. Now we will see how some of this energy may be captured and put to use.

CONDENSER INPUT FILTERS

Suppose we connect a condenser C across $R_{\rm L}$ in the half-wave rectifier



FIG. 15. A condenser alone forms a partial filter, delivering a much higher d.c. voltage providing the current demand is low.

circuit shown in Fig. 15. What will happen? As before, the rectifier will try to produce an output wave form like that shown in Fig. 16A. Notice we say "try"—actually something entirely different will be produced.

When the a.c. voltage makes terminal x positive with respect to terminal y of transformer secondary S, the tube will pass current. As the tube current increases, condenser C is charged up until the condenser voltage builds up approximately to the peak * of the a.c. voltage.

After reaching its peak value, the a.c. voltage begins to decrease and the plate current drops. Now the con-

^{*} The peak value is the maximum positive or negative value of an a.c. cycle. It is 1.41 times the r.m.s. value which is read by an a.c. voltmeter. Thus, the peak is 423 volts when the r.m.s. value is 300 volts.

denser is charged to a high voltage. This charge cannot escape quickly, so when the source voltage drops, it leaves the condenser voltage higher than the source voltage. The condenser voltage is in series with the supply voltage in winding S, between the tube cathode and plate, and (because of the polarity of the condenser) opposes this supply voltage. As soon as the condenser voltage is greater than the supply voltage, the plate of the tube is made negative. This cuts off the current flow through the tube.

The electrons stored in the condenser now leak off through the load $R_{\rm L}$. The condenser thus acts as a source of current for the load, and gradually discharges in the process. When the a.c. voltage again becomes positive, the tube conducts as soon as the supply voltage becomes higher than the voltage of the partly discharged condenser. The condenser is then charged again to the peak voltage of the supply, and the cycle already described repeats.

This action is shown in Fig. 16B, in which the heavy line shows the voltage impressed across the load. As the a.c. voltage builds up from zero to the peak value, the condenser charges. When the a.c. decreases from the peak toward zero, it drops below the condenser voltage, and the tube no longer conducts. Gradual discharge of the condenser brings its voltage down by the time the next pulse rises from zero. Then, the condenser is recharged and current comes from the tube. As the source voltage goes from the peak toward zero, the condenser again takes over.

The important thing for you to understand is that the condenser acts as a source for the load for a longer part of the cycle than does the tube.

The gradual reduction in voltage .across the condenser, shown by the

slanting lines between the peaks, is caused by the discharge of the condenser through load $R_{\rm L}$. If the load resistance is made lower in ohmic value, the condenser will lose its charge and voltage more rapidly. When the condenser is charging, onlythe tube resistance and the transformer impedance limit the charging current, so the condenser reaches full charge almost at once.

The time in seconds it takes to drain a condenser of 63% of its original 1



FIG. 16. The rectifier output when using a condenser input filter.

charge equals the condenser capacity in mfd. multiplied by the load resistance in megohms. Thus, if $R_{\rm L}$ in Fig. 15 has a resistance of 5000 ohms (.005 megohm) and C a capacity of 20 mfd., it will take .005 \times 20 or 1/10 of a second to drain the condenser. But the condenser is being recharged 60 times a second (if the power line is a 60-cycle supply) and therefore never fully discharges. The wave form of the voltage across the condenser is shown in Fig. 16C. This wave consists of a d.c. component and small a.c. ripple components, just like other pulsating currents. But the d.c. average is almost equal to the peak of the a.c. cycle-instead of being only about 3/10 of this value as it was for the ordinary half-wave rectifier shown in Fig. 7B. Clearly, the use of this condenser gives a much higher d.c. output for the same a.c. voltage input,

Furthermore, the a.c. ripple is far less, so the condenser by itself is a partial filter. A regular L-C filter is used between the condenser and load to remove the remaining ripple. A typical power pack using this filter system is shown in Fig. 17. This filter is known as the "condenser input" type, because the tube output goes to C_1 first. You will find condenser input filters in most radio receivers.

▶ But there is a price to pay for the increased d.c. output and lowered ripple obtained from the condenser input filter. The rectifier tube only supplies current for a small period of time, yet this current, by charging the condenser, must operate the receiver continuously. This means the current peaks must be very high and sharp.

The current into the condenser and load from the tube is shown in Fig. 16D. The average d.c. is low, because the rectified current pulses exist for short time periods, yet it must equal the current required by the load.

Therefore, the rectifier tube works very hard for short periods of time. It is important that the load be kept at a reasonable value; otherwise, this terrific rush of current can damage the tube. The tube must be rated for not only the average current but also this peak current.

Where the load resistance is so low that the condenser discharges considerably, the peak current required may be too high for any standard tube. If so, a choke input filter must be used. You will find choke input filters in the power supplies of transmitters, in many public address systems, and in some radio receivers where high power is required.

PRACTICAL FILTER FACTS

The performances of the two main types of filters—choke input and condenser input—are most easily compared by means of curves. Fig. 18 shows a family of curves for a type 5V4G tube using an input filter condenser.

These curves show you how the voltage at the *input* of the filter varies with load current for different input capacitances. (The *output* of the filter will be equal to the input to the filter minus the drop in the filter choke. Since this drop depends on



FIG. 17. A complete condenser input filter.

the d.c. resistance of the choke, and therefore will probably be different for every filter, the best way to compare performances of two different filters is to measure their *input* voltages. This is what we do in Figs. 18 and 19.)

To use the graph in Fig. 18, locate the curve you want. The one at the top marked 400 RMS Volts Per Plate. C = 8 MFD, will do as an example. 400 RMS Volts Per Plate means that this curve was made with 400 volts a.c. applied to each plate of the fullwave rectifier tube. The input condenser has a value of 8 mfd., shown by the notation C = 8 MFD. Now let us see how much input voltage to the filter is obtained when the load is 40 ma. Locate 40 ma. at the bottom of the graph, move straight up to the curve and then straight over to the left of the graph. This brings us to about 538 volts, the input voltage to

the filter when 40 ma. are drawn by the load.

We started with 400 volts and we are getting over 500! But the 400 volts is an r.m.s. value, so the a.c. *peak* is about 564 volts. Our value of 538 volts d.c. is thus close-to the-peak value.

Using the same curve, you will find the input voltage drops to 490 volts at a current of 120 ma. This shows that "loading" the circuit (increasing the load current) reduces the input voltage considerably. This drop is caused mostly by the fact that the condenser had to lose a great deal of charge to supply the higher current. If a smaller input condenser is used, the drop with increased load current becomes even greater (as shown by the curve marked 4 mfd), because the condenser loses its charge even more rapidly.

▶ Now let's use the curves in Fig. 19 to compare the effect of load current changes on the filter input voltage for both choke and condenser input. The condenser curves are the solid lines; the choke curves are dotted.

You see at once that the d.c. voltage at the input of the filter is considerably less for choke input than for condenser input. This is shown by the fact that the condenser input curves for a definite a.c. plate voltage are higher upon the graph. Thus, at a current of 100 ma., we can get about 455 volts d.c. from a 400-volt a.c. input to a condenser input filter, while we can get only about 330 volts when this same 400 volts a.c. is used with a choke input filter.

To put this another way—with a condenser input filter, 400 volts a.c. must be applied to each rectifier plate to obtain 400 volts d.c. with a load of 180 ma. If you use a choke input, you will need about 500 volts a.c. per plate to get the same results. Thus, for a fixed d.c. filter input, a choke input filter needs a transformer capable of producing a higher a.c. rectifier plate voltage than is needed with a condenser input filter.

▶ Should the input filter condenser open up (lose its capacity), the d.c. voltage would be greatly reduced, because the circuit then acts effectively as a choke input filter (as if the condenser were not there.)

▶ The condenser input filter thus has advantage over the choke input type. However, while the voltage drops rapidly as the load current increases in a condenser input filter, it changes very little when a choke input is used. This change in voltage as the load current changes is called regulation.* The better regulation obtained with a choke input makes this arrangement especially valuable when the load current goes through wide variations. Such variations occur particularly in transmitters and in high-power audio amplifiers used in some receivers and in public address systems.

SWINGING CHOKE

The dotted line curves shown in Fig. 19 were made with an input choke having a definite inductance in henrys. If some other value choke is used, the shape of the curve will be approximately the same, but it will be higher up on the graph if the inductance is less, and lower (reducing the filter input voltage) if its inductance is greater.

In other words, the smaller the value of the input choke in henrys, the greater the input voltage to the filter, because this permits the following condenser to act more like an input condenser. As you've just learned, a condenser input filter always has a

^{*} Briefly, regulation is the change in voltage which occurs when the load current changes. The regulation is said to be good when the change in voltage is small.



FIG. 18. These curves show that larger input filter condensers give higher output voltages, for large load currents.

higher input voltage than a choke input filter for the same rectifier plate voltage.

Now, as you learned a short while ago, the *output* of a filter (that is, the voltage furnished to the load) equals the *input* to the filter minus the d.c. voltage drop within the filter. Since the resistance of the filter is constant, this d.c. voltage drop in the filter naturally rises as more current is drawn through the filter. Also, there are d.c. voltage drops in the rectifier tube and transformer. This means that the output voltage of an ordinary filter will drop as more current is drawn by the load.

But if the input choke inductance can be made variable, so it can be decreased when the load current increases, the input voltage will rise and make up for the d.c. drops inside the power pack. Thus, the *output* voltage will remain almost constant.

By using a choke of special design, we can obtain a variable choke inductance. The air gap of this choke is purposely made small so that the core may be easily saturated. As you learned from your study of iron-core coils, saturation occurs when increases in current no longer cause increases in flux. When this condition is approached, the inductance of the choke decreases—which is just what we want to maintain constant output voltage.

Thus, by designing the choke so it will approach core saturation as the load current increases, we get a variable input choke inductance which gives us a more constant load voltage. Since the inductance of such a choke increases and decreases, it is called a *swinging choke*. You won't often find one in radio receivers, but they are common in public address systems, transmitters and electronic control devices.

Of course, the swinging choke gives a variable amount of ripple reduction, so it is used only as an input choke. Another standard filter section is always used after this choke, forming a circuit like Fig. 13.

Saturation will occur in any ironcore choke if d.c. in excess of its rated value flows through it. Leakage in an output filter condenser (such as C_2 in



FIG. 19. Comparing the output of choke and condenser input filters, we find the condenser input type delivers a higher d.c. voltage with poorer regulation.

Fig. 17) can let enough current flow through filter choke L to saturate it. The reduced inductance of L destroys

the filter action and permits the ripple voltage to pass on to the load, causing hum.

Dividing the Voltage

The high d.c. voltage delivered by the power pack filter is used to supply B and C voltages to various tubes. Often these tubes require different plate, screen grid, suppressor grid, cathode and other voltages. Let us now see how these voltages are obtained from the output of the filter.

D.C. Circuit Laws to Remember. In order to understand d.c. voltage distribution circuits, you must remember these important circuit facts:

▶ 1. The currents flowing to a terminal equal the currents flowing away from that terminal (Kirchhoff's Current Law).

▶ 2. In any complete circuit, the supply voltage equals the sum of all the voltage drops (Kirchhoff's Voltage Law).

▶ 3. The voltage drop across any device equals the current times the resistance (Ohm's Law).

 \blacktriangleright 4. The polarity of a voltage drop can be determined by remembering that the terminal of a device at which electrons enter is negative with respect to the terminal at which they leave.

▶ 5. Points connected together by a resistor, coil, or wire through which d.c. current is not flowing are at the same d.c. potential—there is no voltage drop between them.

▶ 6. "Voltage" always means the voltage difference between two points. Thus, when you speak of plate voltage you mean the voltage difference between the plate and cathode, and when you speak of grid voltage you mean the voltage difference between the grid and cathode.

▶ 7. A point can be negative with respect to one point and at the same time be positive with respect to some other. Thus, in a tube circuit, the cathode is negative with respect to the plate and, at the same time, positive with respect to the control grid. ▶ 8. If two points are grounded, there is no d.c. voltage difference between them. They act as if they were wired together directly.

BASIC DIVIDERS

If the power supply furnishes exactly the right amount of voltage for the load, we can connect the load directly across the d.c. supply terminals. Since the voltage is correct, the right current will flow. We need only to be sure the supply can furnish all the current required.

But in any radio device there will be many loads requiring only fractional parts of the supply voltage, and not all of these loads need the same amount of voltage. How can we divide the power supply voltage to meet all requirements?

Series Voltage Drops. If the power source furnishes more voltage than is needed by the load, the correct load voltage may be obtained by placing a resistor between the power supply and load, as shown in Fig. 20A. The supply voltage will divide between the load resistor R_L and the voltage-dropping resistor R_1 . If we choose the correct value of R_1 , just enough voltage will be dropped across it to leave the right amount of voltage for R_L . We can figure the value of R_1 by using Ohm's Law, which says that $R_1 = V \div I$. Here I is the current which is to flow through the load * and V is the excess voltage we wish to drop across R_1 .

Remember—the series dropping resistor may be placed on either side of the load or, if necessary, we can split the series resistor, placing part on each side of the load. The same current flows through each part of a series circuit, so the same voltage drops will occur as long as the total resistance is the same.

Suppose we connect a number of similar loads in parallel, as in Fig. 20B. Now the current through R_1 is the sum of the load currents, so the value of R_1 must be figured by dividing the voltage drop which is to occur



PIG. 20. Series resistors used to drop<u>I</u>the voltage.

across it by the sum of the load currents. This connection can be used when the *voltages* required by the loads are the same—even though different currents may be required.

▶ When different voltages are needed —for example, to supply a plate and a screen grid—the method shown in Fig. 20C may be used. Resistors R_1 and R_2 are figured for the load each is used with.

Both Fig. 20B and 20C can be extended to any number of loads. This voltage dividing method is known as "series voltage dropping," because the dropping resistor is in series and is the only extra part used. \checkmark

Bleeders. Frequently, the load current varies over wide limits. This produces a varying voltage drop across the series resistor, thus changing the division of voltage between the resistor and the load. We can reduce this variation or eliminate it entirely by drawing off or, as technicians say, "bleeding" extra current through the series resistor. A resistor known as a "bleeder" is connected across the load to draw the extra current.

The circuit looks like Fig. 20B, in which R_{L2} will act as a bleeder across R_{L1} . By choosing bleeder R_{L2} so it draws far more current than does R_{L1} , the voltage drop across R_1 will be caused mainly by the bleeder current flowing through it. Then, small variations in the value of R_{L1} can cause only very little change in the total current through R_1 , so the voltage drop across R_1 remains practically This means the voltage constant. across R_{L1} is practically constant no matter what current R_{L1} draws (within limits, of course), so the voltage regu-The lation at R_{L1} is very good. bleeder resistor must be designed to draw a great deal more current than the variations expected in the load current, yet not draw more current than the supply can furnish. Of course, if the load current does not change or the change does not matter, no bleeder is needed—a series dropping resistor is sufficient.

Voltage Dividers. Instead of using a separate series resistor and a separate bleeder resistor, we can use a single resistor, tapped to form a voltage divider as shown in Fig. 21A. Section R_1 is the series section, while R_2 is the bleeder section.

Additional taps may be provided on R_1 to supply various loads (see Fig.

^{*} Since this is a series circuit, the load current also flows through R_{I} .

21B). Here the bleeder current $I_{\rm B}$ through bleeder resistor R_2 flows through all sections of the voltage divider. In section R_4 , currents $I_{\rm B}$ + I_3 flow; in R_3 , currents $I_{\rm B}$ + I_3 + I_2 flow, and in R_1 , we have $I_{\rm B}$ + I_3 + I_2 + I_1 flowing. In each section, the value of the resistor in ohms equals the voltage across the resistor divided by the current flowing through it.

All possible combinations of loads, series dropping resistors and bleeders have not been shown, but Figs. 20 and 21 show the basic methods of voltage division. Reference to them will help you solve any power supply voltage divider network, provided you understand Kirchhoff's Current and Voltage Laws.

WE HAVE STRESSED THESE LAWS BEFORE, BUT THEY ARE SO IMPORTANT IN PRACTICAL RADIO WORK THAT WE RE-PEAT THEM AGAIN:

1. The sum of the currents flowing to a point is equal to the current flowing away from that point. Thus, in Fig. 21B, $I_B + I_3$ equals the current flowing through R_4 .

2. The sum of the voltage drops in a complete circuit equals the source voltage which produced them. For example, in Fig. 21A the supply voltage divides between \mathbf{R}_1 and the parallel resistors \mathbf{R}_2 and **R**_L. The same voltage exists across R_2 and R_L , and this voltage added to that across \mathbf{R}_1 equals the source voltage S. Similarly, in Fig. 21B, the sum of the voltages across R_1 and \mathbf{R}_{L1} equals the source voltage. Also, the sum of the voltages across R_{L2} , R_3 , and R_1 equals the source, and so on.

PRACTICAL DIVIDER CIRCUITS

Fig. 22 gives an example of series dropping resistors in a practical cir-

cuit. Plate current flowing through resistor R causes a voltage drop across R which reduces the supply voltage to the correct amount for the plate. In the same way, screen current through R_1 causes a voltage drop across this resistor which reduces the voltage to the desired screen value. (The resistance of each resistor, as you know, is calculated by dividing the required voltage drop by the electrode current through the resistor.)

▶ The cathode current (which is the sum of the plate and screen currents)



FIG. 21. Examples of voltage dividers.

flows through R_2 , dropping only a few volts across it because of the low ohmic value of R_2 . The polarity* of the voltage drop is as shown. Since the grid connects to the negative side of the drop and the cathode to the positive side, this voltage is between the grid and cathode, and so biases the tube. For this reason, R_2 is known as the cathode resistor, or as a C bias resistor, or simply as a bias resistor.

^{*} To determine the polarity of a voltage drop. remember: THE END OF A RESISTOR (OR OTHER PART) AT WHICH ELECTRONS ENTER IS AL-WAYS NEGATIVE WITH RESPECT TO THE END AT WHICH THEY LEAVE. Tubes serve as signposts for electron direction, since electrons always move from cathode to plate.

▶ If a number of tubes require the same screen voltages, the screens are joined together and the combined screen currents flow through R_1 . Since the combined currents through R_1 will now be greater, the ohmic value of R_1 must be decreased if the same screen voltage is to be maintained. The value of R_1 in ohms will be equal to the required voltage drop divided by the combined currents flowing through it.

Where several tubes require equal C voltages, the cathodes could be joined together. The ohmic value of R_2 would then equal the desired C voltage divided by the combined cathode currents. Ordinarily, however, separate C bias resistors are used for each tube to prevent a possible variation in the current of one tube from affecting the bias of the others.

► To produce equal plate voltages for several tubes, we connect the plate supply circuits in parallel. Resistor



FIG. 22. A practical example showing series voltage-dropping resistors.

R is then figured by dividing the voltage drop required across R by the sum of the plate currents which flow through it. Often the voltage from the power supply is correct for the tube plates; then resistor R is not needed.

▶ If there are other tubes which do not require the same plate, screen and grid voltages, each of these tubes will have its own plate, screen and bias resistors. Even where several tubes use the same voltages, each one may have its own series resistor to prevent variations in one tube from affecting the operating voltages of the others. Fig. 23 shows a typical bleeder voltage divider. Screen grid voltage reduction is produced by R_1 , which carries both the bleeder current drawn by R_3 and the screen grid current of the tube. The bleeder serves to stabilize or hold steady the screen grid voltage. Should the bleeder open, however, only the screen grid current would flow through R_1 , so its voltage drop would be less and more





voltage would be applied to the screen grid.

A practical application of the voltage divider is shown in Fig. 24. Here, resistor R_2 is the bleeder. Series resistor R_3 provides the correct screen voltage, and the C bias voltage is produced across R_1 . Notice that both point 2 on the voltage divider and the cathode of the tube are grounded. The cathode and bleeder current electrons come from the minus supply terminal and flow through R_1 to point 2. Here they divide, the cathode electrons flowing through the chassis to the cathode, and the bleeder electrons continuing on through R_2 .

▶ Fig. 25 shows three common ways of getting bias voltage from the power pack. In Fig. 25A, resistors R_1 and R_2 (this may be a single tapped resistor) are placed in the power supply B— lead. Their action is similar to that of R_1 in Fig. 24, except that the cathode currents of all the tubes now flow through the resistors. Grids of tubes which need the full voltage for bias are connected to C-- while grids of tubes needing less bias are connected to C-.

You recall that a d.c. voltage drop exists across the choke and that the choke can be in the negative side of the circuit. Fig. 25B shows how we can use this drop to supply bias. Here, the drop across the inductance is higher than needed for bias, so the inductance is tapped. This is common practice when the speaker field* furnishes the filter inductance, as the drop may be 100 volts or so and the



FIG, 24. A complete voltage divider.

required bias is usually 50 volts or less. This scheme is mostly used for power output tube bias.

Another way of using the voltage drop across the inductance is shown in Fig. 25C. Here a voltage divider is arranged across the inductance, with the resistances figured to give the desired voltage between B— and each tap. Since no current is taken by grid circuits, these resistors can be high values—in fact, they must be to prevent the resistors from interfering with the action of the inductance.

VOLTAGE REGULATOR TUBES

Sometimes the voltage must be held very closely to the design value. Bleeder resistors are not perfect—in order to hold the voltage within small limits, the bleeder must draw a current many times higher than the possible variations drawn by the load. Often the power supply cannot furnish such high currents.

Further, if a line voltage change makes the power pack output change, the bleeder is helpless. As you can readily see by examining any of the bleeder circuits shown, the voltage across the bleeder will rise or fall as the source voltage rises or falls.

For these reasons, a voltage regulator tube (Fig. 26) is better than a bleeder for keeping voltage constant. This regulator tube VR is a gas-filled tube without a filament. When suf-



FIG. 25. Three ways of getting bias from the power pack.

^{*} The speaker field is a large inductance, requiring a d.c. current flow to establish a fixed magnetic field for proper loudspeaker operation. Since it has sufficient inductance, it can be used as the choke coil in a filter. The normal d.c. current of the set then provides the necessary field excitation. The d.c. resistance of a field is higher than that of the average choke coil, but proper power transformer design makes up for the extra d.c. voltage drop. The method described here makes use of this drop for bias purposes, so the single part is used as a speaker field, a choke coil and a source of bias.

ficient voltage is applied to the tube, the gas ionizes and becomes conductive.

Unlike a bleeder, this tube does not have constant resistance. Its internal resistance depends upon the number of gas ions between the elements of the tube. When the voltage from the filter rises, the number of ions increases; the tube resistance drops, and the tube draws more current. When the voltage drops, the ionization is less; the tube resistance rises, and the current flow drops.

This means that a rise in the filter voltage will cause an increased current flow through VR, which in turn means a larger drop across R, as this current flows through R.

This drop across R automatically compensates for the increase in source voltage, so the voltage applied to tube VT_1 remains very nearly constant. Similarly, when the load current decreases, the IR drop decreases and the voltage across VR increases. This causes a sharply increased current through VR, restoring the drop across R to about its original value.

Should the opposite condition occur, then VR draws much less current, allowing the voltage to rise.

With a resistor R of proper size, this regulating circuit will keep the



FIG. 26. Using a voltage regulator tube.

voltage supplied to VT_1 from varying more than 1 or 2 volts under normal circumstances. The current taken by the VR tube ranges from a few ma. to about 30 ma., which is a reasonably small amount of current.

Voltage regulator tubes rated at 75, 90 and 150 volts are available, known as the VR75, VR90 and VR150 types. By using several of these tubes in series, or by using a voltage divider across a tube, higher and lower voltages can be controlled.

Facts About Rectifier Tubes

Not every radio receiver or amplifier uses the same type of rectifier tube. Just as tube manufacturers make many tubes with different characteristics which can be used for detectors, amplifiers and power output purposes, so do they make many types of rectifier tubes from which the radio designer can choose the one best suited to a particular power supply.

VACUUM RECTIFIERS

Ordinary rectifier tubes are evacuated just like other receiving and transmitting tubes. Hence, they are called "vacuum rectifiers."

You have already seen curves in Figs. 18 and 19 showing the input voltage available from choke and condenser input filters. These curves were taken with a particular type of rectifier tube; they may be quite different for another type, because of the resistance between the plate and cathode of the rectifier tube. This resistance is between the transformer source and the load. When the tube is supplying current, a voltage drop exists in it. Variations in the load current requirement cause a varying tube voltage drop-and hence a variation in the output voltage, since the tube drop

is subtracted from the source voltage.

If the cathode and plate are spaced close together, the tube resistance will be low, and the voltage drop across the plate-cathode of the tube will be low even with high load currents. Naturally, even a fairly large current variation can't cause much total change in the voltage across a low resistance, so this tube construction gives improved regulation. But the a.c. voltage which can be applied to the plate of such a tube is limited. since too much voltage will cause a flash-over between the plate and cathode and ruin the tube. Let's look into this matter of flash-over a little further.

Peak Inverse Voltage. A rectifier tube should pass electrons from the cathode to the plate when the plate is positive. But what happens when the cathode is positive with respect to the plate, a condition which exists once every cycle of the a.c. supply?

If the voltage is too high, a current will arc (spark) between the elements or between the leads coming through the glass "pinch" (or seal) of the tube. This either breaks the seal, letting air into the tube, or destroys the emission qualities of the cathode. The tube is ruined in either case.

The maximum a.c. voltage that you can connect to a rectifier tube without causing an arc when the plate is negative and the cathode is positive is called the *peak inverse voltage* rating of the tube.

Fig. 27 shows a single diode connected through a load R to a highvoltage secondary. Once each half cycle the tube plate is positive (+), while the cathode (K) is negative (-), and once each half cycle the plate is negative (-) and the cathode is positive (+). When the diode passes electrons, the voltage across the tube is lower than $V_{a.c.}$ because of the voltage drop in the load R. But when the inverse condition exists (that is, when the plate is negative), no current flows through the tube. Then the voltage $V_{d.c.}$ stored in the condenser C, added to the peak $V_{a.c.}$ across the high-voltage secondary, is applied between the cathode and plate. If the load current is low, the voltage across C may nearly equal the peak of $V_{\rm a.c.}$ (The peak value is 1.41 times the value measured by an a.c. voltmeter.) Therefore, the maximum inverse voltage which is applied between the plate and cathode is $V_{a.c.} \times 1.41 \times 2$ twice the peak voltage of the transformer secondary. This value must not exceed the peak inverse voltage rating of the tube.

In the case of a full-wave rectifier, the peak inverse voltage will be applied first between one plate and its cathode, and then between the other plate and its cathode. Therefore,



FIG. 27. The peak inverse voltage is the sum of the d.c. output and the peak a.c. input.

both diodes of the full-wave rectifier tube must have the same peak inverse voltage rating.

With a choke input filter in place of the condenser, the voltage $V_{d.c.}$ is considerably reduced. This means that when an input choke is used, a larger a.c. voltage may be applied to the rectifier plates without danger of exceeding the manufacturer's inverse peak voltage rating.

Thus, in choosing a rectifier tube, we must consider not only its d.c. output current and voltage, but also its peak inverse voltage rating and the peak current which it can safely pass.

MERCURY VAPOR RECTIFIERS

Where we need both high voltage and good regulation, we cannot use a vacuum type tube because the close spacing necessary for good regulation limits the peak inverse voltage rating. Instead, the mercury vapor tube is used. This tube is like a vacuum type with wide plate-to-filament spacing, but contains a small ball of mercury. When the tube heats up, a quantity of mercury vapor is formed and at the same time, an electron cloud forms around the filament.

As electrons speed toward the plate, they strike mercury atoms in the vapor and knock out free electrons, leaving positive gas ions. The extra electrons move toward the plate along with the original electrons. The positive ions move toward the filament getting into the electron cloud, where they combine with electrons and partially destroy space charge effects. This lets more electrons reach the plate from the filament, because fewer are repelled by the cloud back to the filament.

When the load requires more current, more electrons pass through the tube. This increases the collisions with mercury atoms, thus producing more free electrons and further reducing the space charge. Effectively, this is the same as reducing the plate-tocathode resistance. The increased current through the smaller tube resistance gives about the same tube voltage drop as before, so the output voltage remains almost constant. In fact, the tube drop stays at about 15 volts for all values of current which the tube can safely handle. This constant voltage drop means the d.c. output does

not vary with load current changes, so regulation is excellent.

The heavy ions must not strike the filament as they can destroy its emission. Hence, the peak current must be kept within the tube rating and the electron cloud must form before the tube passes current. Since the mercury vapor tubes used in receivers have quick-heating filaments, this is no problem. When large mercury vapor rectifier tubes are used in transmitters and high power public address systems, a special time-delay relay is



FIG. 28. The mercury vapor tube requires r.f. chokes in the plate leads to keep down interference.

used to turn on the plate power after the filament power is applied and the tube has warmed up.

A typical full-wave mercury vapor rectifier circuit is shown in Fig. 28. In operation, the rectifier tube emits a blue glow which quickly distinguishes it from the vacuum type rectifier tube (which emits no glow when in good condition*). The filter circuit has a choke input, for a condenser

^{*} When a vacuum tube rectifier has been overloaded by excess current, usually caused by a shorted filter condenser, gas is often formed in the envelope. Then there will be a blue glow which at first can be seen inside the space between the plate and cathode. Later, this glow will increase as more gas molecules are released from the metallic structure of the tube. The glow may spread over the entire interior of the tube. Frequently, the excess current causes the rectifier plates to become red hot. The glow in a vacuum type tube is steady and is not usually as brilliant as the flickering glow of a mercury vapor rectifier.

input would cause sharp peak current rushes which would ruin the tube. To improve regulation, the input choke is usually a swinging choke.

Sudden current changes in the rectifier circuit may set up radio frequency (high-frequency a.c.) ourrents which will interfere with the signals in the signal circuits. For this reason, small r.f. chokes RFC are used in each plate circuit. The tube may have to be shielded to prevent stray fields from getting into the signal circuits. Shielding should not be used unless necessary, as it prevents the heat generated by the tube from escaping and so eventually causes tube destruction. A fuse in the power line is recommended; then, if a short circuit or sudden overload causes excessive tube current, the fuse will burn out before the tube cathode is damaged.

BRIDGE RECTIFIER CIRCUIT

The bridge rectifier circuit is a circuit which uses four diodes to get the effect of full-wave rectification from a half-wave transformer. Since the saving in the cost of the transformer is more than the tube cost only in high-power equipment, bridge rectifier circuits are usually found only in transmitting and measuring circuits. Full-wave rectifiers are much more common in radio sets.

A typical bridge circuit is shown in Fig. 29. You will find it easy to remember this circuit by following the electron flow. For example, when terminal x is negative with respect to terminal y of transformer secondary S_1 , electrons flow from the supply terminal x of S_1 to the rectifier terminal a, and seek a path through the bridge circuit to terminal y. Since the only electron path through a normal tube is from the cathode to the plate, the

electrons cannot go through tube VT_4 ; instead, they pass through tube VT_1 to point b. The only electron path from point b is through the load resistor R to point c, because the path through tube VT_2 is blocked. From point c, the electrons travel through tube VT_s to point d and then to the supply terminal y. This electron path is shown by the solid black arrows.

When the cycle reverses, electrons leave the supply terminal y and take the $y \rightarrow d \rightarrow b \rightarrow c \rightarrow a \rightarrow x$ path, indicated by the white arrows.



FIG. 29. A bridge rectifier circuit.

As you see, electrons flow from point b to point c no matter which terminal of the supply they leave. Thus this circuit gives us full-wave rectification between points b and c without a center tap on the transformer and with a transformer source voltage equal to that fed only one plate of the usual full-wave circuit. Points b and c form the output terminals of the rectifier; since electrons flow from b to c, b is the negative (-)and c is the positive (+) terminal.

Of course, a standard filter circuit is connected between the rectifier output terminals and the load. A voltage divider can be used too, if one is needed.

Lesson Questions

Be sure to number your Answer Sheet 12FR-2.

Place your Student Number on every Answer Sheet.

Send in your set of answers for this lesson immediately after you finish them, as instructed in the Study Schedule. This will give you the greatest possible benefit from our speedy personal grading service.

- 1. Into what two parts may the circuits of a radio receiver be automatically divided?
- 2. Briefly give the three important requirements of a power supply.
- 3. What would you expect to happen to a 2.5 volt filament tube, if it were placed in a 6.3 volt tube socket?
- 4. What type of current exists at the output of the rectifier tube: (1) a.c.;
 (2) pure d.c. or (3) pulsating d.c.?
- 5. What two parts are most widely used in an a.c. filter?
- 6. What happens to the d.c. voltage when an input filter condenser opens up (loses its capacity)?
- 7. As far as the filtering action is concerned does it matter whether the filter choke is placed in the positive or the negative side of the circuit?
- 8. Why is extra current drawn through a voltage divider by a bleeder resistor? LOAD STABILIZER,
- 9. Suppose the bleeder resistor opens in the circuit shown in Fig. 23. Would you expect the screen grid voltage to: (a) *increase;* (b) *decrease*, or (c) *remain the same?*
- 10. Would you expect to find: (1)<u>a choke input;</u> or (2)<u>a condenser input</u> filter used with a mercury vapor rectifier tube?

CASHING IN ON DISCONTENT

Discontent is a good thing—if it makes you want to do something worth while. If you had not been discontented, you would never have enrolled for the N.R.I. Course.

Practically every one is discontented. But some of us are "floored" by discontent. We develop into complainers. We find fault with anything and everything. We end up as sour and dismal failures.

Those of us who are wise use our discontent as fuel for endeavor. We keep striving toward a goal we have set for ourselves. We are happy in our work. We face defeat, and we come out the victors.

At this minute you may be discontented with many things—your progress with your Course, your earning ability, yourself.

Make that discontent pay you dividends. Don't let it throw you down. If you do, you may never be able to get up again. Keep striving to remove the cause of your discontent. Remember that it's always darkest before the dawn. And a real N.R.I. man works hardest and accomplishes most when he is face to face with the greatest discouragements.

J. E. SMITH.

SPECIAL POWER SUPPLIES FOR RADIO EQUIPMENT

13FR-2

NATIONAL RADIO INSTITUTE ESTABLISHED 1914 WASHINGTON, D. C.



STUDY SCHEDULE NO. 13

For each study step, read the assigned pages first at your usual speed. Reread slowly one or more times. Finish with one quick reading to fix the important facts firmly in your mind, then answer the Lesson Questions for that step. Study each other step in this same way,

midget receivers. This supply has the advantage of small size and low cost. Also, it will operate from any 110-volt power line, whether a. c. or d. c. Special methods are used to obtain the needed operating voltages, even to the development of special tubes for such receivers. Answer Lesson Questions 1, 2, 3 and 4.

but a power transformer is not wanted, a voltage doubler can be used. These circuits are able to step up the voltage as long as the current demand is low. Both the half-wave and full-wave types are found in some of the larger table model receivers and in other radio and electronic equipment. Voltage doublers will work only from an a. c. power line, as they depend on the phase reversal on alternate half cycles for operation. Answer Lesson Question 5.

- equipment. This unit removed the requirement for bulky, expensive batteries or an expensive motor generator, by making it possible to obtain standard voltages from a single low-voltage battery which is already available in the car. Here you learn how the mechanical switch changes d. c. into a pulsating current which can be stepped up by a transformer and treated like a. c. A tube rectifier can be used, or another vibrator section can be made to convert the high-voltage a. c. to d. c. again. A vibrator power supply comes closest to duplicating the standard a. c. supply, in regard to the voltage and current output. Answer Lesson Question 6.
- A truly portable receiver must run from a self-contained power supply. Batteries are ideal, particularly since modern low-drain tubes operate from small light-weight cells. These low-current tubes have also made it possible to obtain filament power from the ordinary universal power pack, as well as plate and grid voltages. Thus, by using a small a. c.-d. c. power supply as well as batteries, the three two models are been as the tubes of the tubes of the tubes of tubes of the tubes of the tubes of tubes o the three-way portable can be run from power lines when available, thus saving the batteries. Answer Lesson Questions 7, 8 and 9.
- some piece of standard or special equipment. Also, you may be called on to convert from one power supply type to another, due to differences in available power or for some special operation. The methods are discussed here, so you can see just what is possible and the best way of getting the desired operation. Answer Lesson Question 10.
- □ 6. Mail Your Answers for this Lesson to N. R. I. for Grading.
- 7. Start Studying the Next Lesson.

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World Radio History

SPECIAL POWER SUPPLIES FOR RADIO EQUIPMENT

The Universal A. C.-D. C. Power Supply

THE standard a.c. power supply you've studied so far uses a power transformer to step up the line voltage. The stepped-up voltage is then applied to a rectifier tube, which in turn feeds the load through a filter. This standard power supply is nearly always used when high voltages or large currents are needed.

▶ But sometimes it is impossible or undesirable to use a transformer in the power supply. In some sections of our largest cities, for example, the power lines furnish d.c. instead of a.c. Of course, a transformer can't work on d.c.

Then, too, some places have 25-cycle current instead of the usual 60-cycle. This requires a special, more expensive transformer.

There are also cases where a power transformer is unnecessary—modern midget receivers, for example. These sets will operate reasonably well on just the line voltage. Further, they are so small that there is no room in them for a bulky transformer. And finally, they are built to sell at very low cost—which is possible only if the expense of a power transformer is eliminated.

▶ Now we are going to study a universal a.c.-d.c. power supply which contains no power transformer. It is designed to operate from any a.c. or d.c. power line, and is entirely satisfactory for small sets and low-power apparatus. Let's first see how the B and C supplies are obtained, then go into the question of filament supply a little farther on.

The B-C Power Pack

A simple universal power supply is shown in Fig. 1. As you see, no power transformer is used—the power line connects directly to the circuit. The half-wave rectifier tube VT is followed by a standard condenser input filter which connects to the load.

This circuit works just like the standard half-wave rectifier with a condenser-input filter that you've already studied. That is, when the source voltage makes point 2 positive with respect to point 1, the tube conducts; electrons flow from point 1 through load $R_{\rm L}$, and $C_{\rm l}$ is charged to the polarity shown. When the voltage reverses, making point 1 positive with respect to point 2, the tube no longer



FIG. 1. This universal power supply uses no power transformer.

conducts; C_1 partially discharges through load R_L , maintaining an electron flow through it. Thus, the only difference between this circuit and the standard half-wave rectifier circuit is that we've managed to eliminate the power transformer. (Remember, the circuit is completed between points 1 and 2 by the power line leading to a generator.)

Of course, the maximum possible voltage across the load in this circuit is the peak value of the power line voltage. As most power lines deliver about 115 volts,* the <u>peak value</u> will be about 165 volts. Actually, when current is taken from the power pack, the drop in the tube usually brings the total voltage across the load down to about 100 volts. Obviously, from what you know about filters, a condenser-input filter is necessary in a universal receiver to get a higher-output voltage than would be delivered by a choke-input filter.

As in the standard power supply, the choke coil in Fig. 1 can be placed at point X (between the negative leads of the filter condensers), and the voltage drop across it can then be used to furnish bias voltages.

Circuit Variations. Many receivers use dynamic loudspeakers, which have a field coil that must be energized by d.c. from the power sup-



FIG. 2. The speaker field is energized from a separate diode here.

ply, in order for the speaker to operate. If the coil has a low resistance (500 ohms or less), we can use it in place of the filter choke in Fig. 1 and thus make the coil do double duty.

▶ But most speakers used in a.c.-d.c. sets have a field coil resistance of around 2500 ohms. A coil with such high resistance can't be used as a filter choke in our universal circuit, because there would be too much d.c. voltage dropped across it, and therefore too little output from the filter. When we use such a speaker, we have to put a regular choke in our filter section, and find some other way of connecting the speaker coil to the power supply to give the coil the current it needs.

A circuit which does this is shown in Fig. 2. Here, a dual diode tube is employed. (Separate diodes could be used, if desired.) One diode is used as a half-wave rectifier for the load R_L ; the other acts as a rectifier to supply the speaker field, marked L_1 , which acts as the load for this tube section. The filter condenser C_8 provides sufficient filtering for the field. The other filter C_1 -L- C_2 is a standard condenser-input filter, used to deliver pure d.c. to the load R_L .

▶ As you know, the input filter condenser draws a rather high current from the rectifier tube until the condenser is fully charged. As a result, in the circuit shown in Fig. 2, each of the rectifier tube sections undergoes considerable strain when the circuit is first turned on. Fig. 3 shows how the circuit can be changed to relieve some of this strain and save one filter condenser at the same time.

Here the two cathodes of the rectifier tube sections are tied together, as are the plates. This lets us use the current capacity of the two diodes (which are now in parallel) to charge condenser C_1 , thus reducing the current each tube section must carry. The speaker field L_1 , in parallel with C_1 , is connected directly across the output of the rectifier. Enough filtering is produced by C_1 for the field supply. C_1 thus takes the place of condenser C_3 in Fig. 2, and at the same time acts as the input filter condenser for the filter C_1 -L- C_2 .

▶ In very inexpensive receivers, the filter choke may be replaced by a

^{*} Standard power lines are rated at 110 volts, 115 volts or 117 volts. Actually, these ratings mean the same kind of power line, as the power line voltage may be any value between 100 volts and 120 volts, depending on the load on the line. The peak value is about 1.4 times the r.m.s. value for which the line is rated.

cheaper. but less efficient resistor. Fig. 4 shows a typical circuit. If the resistor R_1 could be made large enough so that the ratio of its resistance to the reactance of condenser C_2 was about the same as that of the choke reactance to condenser reactance, the filtering obtained would be approximately the same. However, a resistor this large cannot be used, because the d.c. voltage drop across it would be so great that little voltage would be left for the load. A smaller resistor must be used, hence, the filtering is not as good as that obtained with the average choke.

The resistor can be made larger, improving the filtering, if we can reduce the d.c. current flow through the resistor. As most of the d.c. required is for the plate of the output tube, it seems logical to divert this flow. By connecting the output tube plate circuit between terminal X and B— in Fig. 4, this plate current does not go through R_1 .

This method has the disadvantage that the only filtering given the out-





put tube plate supply is that furnished by C_1 , so a considerable a.c. ripple is fed to the output tube. However, since there is no amplification between the output tube plate and the loudspeaker, and since the output transformer and speaker will seldom respond to low frequencies in these inexpensive receivers, this ripple will not cause an objectionable hum.

If the screen grid of the output tube or any electrode of any other tube is connected to X, too much hum will be produced. Therefore, all the rest of the tube electrodes are connected between B+ and B-, so they will benefit from the additional filtering of R_1 and C_2 .

Regulation. The filter systems of Figs. 1, 2, 3 and 4 are condenser-input filters. As you know, their regulation



FIG. 4. Resistor R_1 is used in place of a choke coil, making a less efficient but still useful filter.

will depend on the amount of current taken by the load and upon the amount that condenser C_1 is discharged in between pulses of current from the rectifier. Just as in the standard power pack, the less the load current demand, or the larger the input condenser, the nearer the output voltage will approach the peak of the a.c. rectifier plate voltage.

Fig. 5 shows the regulation obtained with different sizes of filter condensers. You will notice that the d.c. voltage is high when low currents are drawn, but drops off very rapidly as the current demand is increased. It is easy to see that the condensers must be high-capacity types to give a reasonable output.

Tube Protection. As you already know, a large input condenser draws a high peak current from the rectifier tube. When a transformer is used, the resistance of the high-voltage winding limits the current through the rectifier tube to some extent; but the power line current in Fig. S is limited only by the fuses in the house wiring, which may have ratings as high as 25 amperes.

This is no protection at all for the rectifier tube. Many of these tubes

were ruined by excess peak current \bullet before receiver manufacturers hit upon the idea of placing a resistor in series with the plates of the tubes. A single resistor of from 20 to 30 ohms may be placed in series with both plates at point *a* of Fig. 3, or a 50-ohm resistor may be placed in series with each plate at points *b* and *c*.

Now, even if condenser C_1 is considerably discharged, the maximum current flow is limited by the series resistor plus the tube resistance to a value less than 2 amperes. Even so, an overload lasting for a considerable time (which might be caused by a defective condenser) will either burn out the protective resistor or destroy the emission of the tube.

Filament Supply

So far, you have learned how to eliminate the power transformer and still obtain about 100 volts for plate and C bias supplies. Now how shall we supply the filaments?

Early a.c.-d.c. receivers used cathode type 6.3-volt a.c. tubes, obtaining the required filament voltage from the power line by using a resistor to drop the line voltage. Let's see how this can be done.

Suppose we have five tubes rated at 6.3 volts. If we connect these tubes in parallel, we will need 6.3 volts to operate them, and the total current demand will be the sum of all the filament currents. Presuming these five tubes have filaments rated at .3 ampere each, a current of 1.5 amperes must be taken from the power line $(5 \times .3 = 1.5)$. If we use a series resistor to drop the voltage from the line voltage value of 115 volts to the desired 6.3 volts, about 108 volts must be dropped in the series resistor. At a



FIG. 5. Regulation curves for the univarsal power supply.

current of 1.5 amperes, the power dissipated in this resistor is about 162 watts. (Watts = current \times voltage, $1.5 \times 108 = 162$ watts.)

This is entirely too much power to waste. We will waste much less by arranging the tube filaments in series, as shown in Fig. 6.

Since the same current flows through all the parts of a series circuit, the tube filaments shown in Fig. θ must have the same current rating. Let's assume again that this is .3 ampere, and that the tubes are 6.3-volt tubes. The voltage drop across the tubes will be the sum of their voltage ratings, so there will be about 31 volts dropped across the tubes. As a result, only about 84 volts must be dropped across the resistor R. At a current of .3 ampere, the power loss in the resistor is now only 25 watts. Obviously, this represents a great saving in power.

As soon as a.c.-d.c. circuits proved

^{*} Excess current through the tube will make the cathode lose its electron-emitting ability. It may even cause the plate to sag and touch the cathode. Usually, however, the metal strip inside the tube which is used to connect the cathode to the lead from the tube base acts somewhat like a fuse, burning out if the current is much greater than 2.5 amperes. This metal strip cannot be made heavy, because it would then conduct too much heat away from the cathode of the tube.

practical, tube manufacturers began to make tubes with higher filament voltage ratings. The ratings jumped to 12, then to 25 volts. In particular, the rectifier and power output tube voltages were increased.

Since the current ratings of these tubes were kept at the same value as the available 6-volt tubes, the new tubes and the 6-volt tubes could be connected in series. As you know, voltage drops add in a series circuit so using tubes with higher filament voltage drops means that less voltage must be dropped by resistor R. Consequently, less power is wasted in this resistor.

When the tubes are in series, resistor R can be an ordinary 35- or 50watt resistor, as long as it has the proper resistance value.* However, a regular 35- to 50-watt resistor is rather bulky; further, dissipating this amount



FIG. 6. The resistor drops the voltage to the amount required by the series-connected filaments.

of heat underneath the chassis may dry up the electrolytic filter condensers.

*You may have to figure the value of resistor R in some repair jobs. To do this, add the voltage drop across the tube filaments. Subtract this sum from 117 volts, which is considered the line voltage value. The remainder is the voltage which is dropped across the resistor R. Now look up one of the tubes in a tube chart to find what the filament current is. Divide the resistor voltage drop by this current value and you will get the resistor will be the resistor voltage drop multiplied by the current through the resistor. To be safe, choose a resistor with a higher wattage rating than this figured value.

Ballast Tubes. To avoid excessive heat under the chassis, the resistor may be placed in a tube known as a ballast tube or resistance tube. This



FIG. 7. Using a line cord resistor. In practical circuits of this type, you may find a protective resistor in series with the rectifier plates; also, there may be a speaker field connected across input filter condenser C₁.

is a metal or glass tube with a regular tube base, which has only a resistance element in it. The resistance element is wired in series with the filaments of the other tubes, so the circuit is the same as Fig. 6. A ballast tube dissipates heat like all other tubes, from the top of the chassis, and so causes no heat under the chassis. Tubes with various resistance values may be obtained to suit the circuit.

Line-Cord Resistors. Another common voltage-dropping resistor is the Cordohm or line-cord resistor. This is a line cord having three (or more) wires. A typical circuit using a Cordohm is shown in Fig 7. Two of the wires form the regular a.c. power cord going to the B power pack. The third wire is a resistance unit of the proper size for the particular tube combination it is used with. This resistor value is figured the same way as an ordinary resistor, when a replacement is necessary. Wattage ratings need not worry you, however, as

these cords can safely dissipate 30 watts.

▶ Since this resistor unit runs the entire length of the cord, the cord will become warm. This may alarm some of your customers; if so, explain to them that the cord contains a resistor which is supposed to dissipate heat outside the set chassis. For this reason, such cords should always be stretched out to their full length, never bundled up inside the radio. Also, these cords should never be cut to shorten them, as this reduces the resistance, even if the proper connections are made.

Eliminating the Series Resistor. In many of the more recent receivers, new types of tubes are used which have filament voltage ratings that add up to the line voltage total. A typical circuit is shown in Fig. 8. Here the 35Z5 rectifier tube requires 35 volts for its filament. The 50L6 output tube requires 50 volts, and the remaining tubes require 12 volts each. (Notice that the first part of the tube number shows the approximate filament voltage needed for the tube. This is standard practice in numbering the newer tubes.) The total is about equal to the line voltage, so no series resistor is needed.

Grounding Precautions

In Figs. 7 and 8, switch SW (the ON-OFF switch of the set) opens the filament and B supply circuits simultaneously by opening one side of the power line. Notice that one end of the switch, one end of the filament string and the B— terminal are shown going to ground symbols. This means that these three terminals are all connected together. For convenience, each point is often connected to the chassis; since the chassis is metal and a good conductor, this is exactly the same electrically as connecting the three wires directly.

Since one wire of the house power line is grounded and this ground is connected to the chassis by the cord of the receiver, no set wired like this should ever be connected to a ground by a separate wire. If it is, and the power cord plug happens to be inserted in the wall socket so the *ungrounded* wire of the house line is connected to



FIG. 8. The tube filaments have ratings adding up to the line voltage, so no series resistor is needed. (The 35Z5 tube filament is shown separately from the diode here for convenience, both are parts of a single tube.)

the chassis, the house line will be shorted through the set. Of course, this will blow out the power line fuses. *NEVER USE A GROUND CON-NECTION ON AN A.C.-D.C. RE-CEIVER* unless a terminal is provided by the set manufacturer. And, for your personal safety, remember that the set chassis will be 115 volts above ground if the plug happens to be in the socket upside down. If it is, you will get a shock if you touch the chassis while you are standing on a concrete floor or other ground.

▶ In many a.c.-d.c. receivers, the chassis is not an electrical part of the circuit. In such sets, points marked with a ground symbol on diagrams are connected by a wire (often called a "bus"). The chassis may be connected to this ground bus through a condenser and resistor (C_1 and R_1 in Fig. 7). The chassis is indicated by the special symbol shown in this figure

when it is not used as a part of the circuit.

Dial Illumination

A pilot lamp is used in most radios to light the dial and show when the receiver is turned on. Let's see how the lamp is wired into the circuit.

The obvious way is to use a pilot lamp with the same current rating as the tube filaments, and connect it in series with them. Unfortunately, this method won't work. When the radio is first turned on, the tube filaments are cold and have lower-than-normal resistance. It takes several seconds for them to warm up. On the other hand, the pilot lamp warms almost immediately. Thus, when the radio is first turned on, the low filament resistance allows a higher-than-normal current to flow for a few seconds which would burn out the pilot lamp.

Instead, we put a small resistor in series with the filaments, and connect the lamp across it. The lamp usually has a lower current rating than the tube filaments. The shunting resistor across the pilot lamp carries the difference in current between the pilot lamp rating and the current flowing in the circuit. For example, if the pilot lamp is rated at .15 ampere, and .3 ampere is flowing, the shunt resistor must carry the difference between the two, or .15 ampere.

The resistor has such a value that the voltage drop across the resistor and lamp is 4.25 volts * when the normal filament current flows. The drop is somewhat higher during the first rush of current, but now does not exceed the 6.3-volt rating of the pilot lamp. This means the lamp does not run at full brilliance except during the filament warm-up period. However, it gives enough light—and will not be burned out by the high initial current.

When a series filament resistor, ballast tube or line cord is used, a tap on the resistor element may provide a shunt resistance, as shown in Fig. ϑ . If desired, the section R_1 can be a separate resistor.

▶ In Fig. 8, the 35Z5 filament in the filament string has a tap on it, so a pilot lamp can be connected across a part of the tube filament, which acts as the resistor element in shunt with the pilot lamp.

Connecting the plate of the 35Z5 to this filament tap, as shown in the diagram, makes it unnecessary to use a protective series plate resistor. This puts the parallel combination of the pilot lamp and the tube filament shunting section in series with the



FIG. 9. The pilot lamp can be connected across a tapped section of the filament resistor.

plate of the rectifier tube; if too much current flows, the pilot lamp will burn Should the overload continue, out. this section of the tube filament may open. But usually, if the set is turned off quickly after the pilot lamp burns out, the pilot lamp will be the only part damaged. Before replacing the bulb, the filter condenser should be . checked to be sure the overload is not the result of a defect in it. Incidentally, in any a.c.-d.c. receiver, a burned-out pilot lamp should be replaced at once. Never operate the receiver with the pilot lamp burned out, because another part may eventually be damaged.

^{*}The resistor value can be found by dividing 4.25 volts by the current it must carry. This current is the difference between the filament current and the pilot lamp rating.

▶ The most common pilot lamps for a.c.-d.c. sets are 6.3-volt bulbs rated at .15 or .25 ampere. The .15-ampere bulb has a brown colored glass bead supporting its filament, while the .25ampere bulb has a blue bead.__Nat-_ urally, you should always replace a lamp with another of the same rating.

▶ You will probably often be asked by owners of a.c.-d.c. sets why the pilot lamp flares up brilliantly when the set is just turned on, then grows considerably dimmer. The brilliance at the start is caused by the initial rush of filament current through the pilot lamp. Then when the tubes warm up, the current through the pilot lamp decreases, making the lamp less brilliant.

The pilot lamp shown in Fig. 8 will go through this same action, then become brighter again. As in Fig. 9, the first brilliant flare is caused by the extra current flow in the filament circuit. The pilot lamp dies down until it practically goes out as the filaments warm up. Then when the tubes draw plate current from the B power supply, the rectifier plate current flows through the pilot lamp and again lights it up.

Operation on D. C. Power Lines

Any a.c.-d.c. receiver can be used on a d.c. power line. All you need do is place the power plug in the wall socket in such a way that the plate of the rectifier tube is positive. Then current will flow continuously through the rectifier tube. The tube just passes on current, and does not act as a rectifier at all: however, it does act as a safety device for the electrolytic condensers. As vou know, an electrolytic condenser will break down and become a low-resistance conductor if voltage of the wrong polarity is impressed across it. The rectifier tube prevents this from happening, because it will not pass current if the power plug is installed with the wrong polarity.

The dotted line in Fig. 5 shows the regulation obtained when d.c. voltage is applied to the input of an a.c.-d.c. receiver. Since the rectifier tube passes the needed d.c. average all the time, the filter condensers do not discharge, and the output voltage is practically independent of the effect of the filter circuit. Instead, it depends primarily on the resistance of the rectifier tube and filter choke. The only use for the filter circuit now is to remove any stray ripple voltages which may be on the d.c. power line.

Voltage Doublers

Sometimes, a higher voltage than is obtainable from the usual universal a.c.-d.c. power pack is wanted, without using an expensive transformer. This is made possible by another transformerless circuit called a voltage doubler. Let us see how it works.

Full-Wave Voltage Doubler

If we can charge two condensers, then put them in series so that their voltages add together, we will get twice the charging voltage from the series combination. This is the theory of the full-wave voltage doubler circuit shown in Fig. 10.

This circuit resembles a half-wave rectifier circuit using a condenser-input filter, except that it has an extra condenser and an extra rectifier. The easiest way to understand its operation is to trace the flow of electrons. When the a.c. cycle makes point 2aegative with respect to point 1, electrons will flow from point 2 to point 1. How do they get there? There are two possible paths, through C_1-VT_1



FIG. 10. This circuit doubles the line voltage without using a transformer.

and through C_2 - VT_2 . However, electrons cannot flow from a plate to a cathode, so the C_2 - VT_2 path from point 2 is blocked when point 2 is negative. Electrons flow instead into one plate of condenser C_1 , and an equal number of electrons are forced out of the other plate. This charges condenser C_1 with the polarity shown. The electrons leaving C_1 flow through VT_1 to point 1. The circuit is shown by the black arrows.

When the a.c. line voltage polarity reverses, electrons leave point 1 and, finding their way blocked by tube VT_1 , pass through VT_2 and charge C_2 . Those electrons driven out of C_2 during the charging process flow to point 2. This half cycle is shown by the outline arrows.

Notice that condensers C_1 and C_2 are alternately charged from the power line. Also notice the polarity of the voltages across the condensers. Since they are in series, the voltage between terminals *a* and *b* is approximately the sum of the condenser voltages. Of course, one condenser is being discharged by the load while the other is charging, so we don't get quite double the peak of the line voltage. The actual voltage depends upon the amount of load and the size of the condensers.

This circuit is known as a full-wave voltage doubler because both halves of the a.c. cycle are rectified. Hence, 120 cycles is the lowest ripple frequency which must be filtered by $L-C_{\rm s}$ if the power line furnishes 60-cycle current.

Fig. 11 shows the regulation characteristics of a full-wave voltage doubler in which condensers C_1 and C_2 are of equal value. Curves are shown for values of 5, 10, 20 and 40 microfarads for each condenser.

You can see that increasing the load current results in considerable voltage drop. In other words, the regulation is poor. The larger the capacity of the condenser, the less the drop in d.c. voltage for increased load current—but even at best, the regulation of this circuit is like that of a universal



FIG. 11. The regulation of a voltage doubler is like that of the universal power supply.

a.c.-d.c. power pack, and is considerably worse than that of a transformer type power pack. It can be used satisfactorily only where the load current is not liable to vary greatly or where high currents are not required.

A Practical Circuit. Of course, Fig. 10 is a simplified circuit. A typical practical circuit, using a dual-
diode tube, is shown in Fig. 12. In its important details it is the same as the circuit shown in Fig. 10.

Let's trace the electron path through Fig. 12. When point 2 (near the power plug) is negative with respect to point 1, the path is from point 2 to point c, through C_1 to cathode K_1 , and from P_1 through resistor R_1 , to point 1. (R_1 and R_2 are protective resistors in series with the tube's plates.) On the next half cycle, the path is from point 1 to cathode K_2 , to P_2 , through R_2 to point b, through condenser C_2 to point c, and thus back to point 2. Therefore, the circuit works exactly like Fig. 10.

In addition, the filament circuits are drawn in. You can trace the cir-



FIG. 12. A practical full-wave voltage doubler.

cuit from point 1 through resistor R_3 , through the filament of the rectifier tube and in turn through tubes VT_3 , VT_2 and VT_1 , back to point 2.

▶ Notice that the cathodes of tubes VT_1 , VT_2 and VT_3 all connect to B_{--} , as is normal in radio receiver circuits, but the filaments are connected directly across the power line. The filament of tube VT_1 , which connects to point 2, also connects to point c (between the filter condensers). Therefore, the voltage difference between the cathode of this tube (which connects to point b) and the filament (which connects to point c) is the

voltage across condenser C_2 . This means there is a d.c. voltage between the cathode and filament of this tube which is approximately equal to the line voltage. Modern tubes have excellent insulation between the filament and cathode, so there is little danger of this voltage breaking down the tube between these elements, but a certain amount of leakage between them may develop. Since the filament of VT_1 is positive with respect to the cathode, it may attract some electrons from the cathode, so there ean be a current flow through the space within the tube between the cathode and filament. The a.c. voltage across the filament will vary this leakage current and thereby create a certain amount of hum. As you will learn later, tube VT_1 is usually the first audio or detector tube, where not even a slight hum can be allowed. Careful tube selection for a minimum amount of leakage is necessary in these cases. Voltages exist between the cathodes and filaments of the other tubes, but leakage currents in these tubes are not so important.

Now let's look into another voltagedoubler circuit, which will solve some of this leakage current difficulty.

The Half-Wave Voltage Doubler

Like the full-wave voltage doubler, the half-wave voltage doubler uses a condenser as a voltage source—but as a source of voltage for a rectifier tube. Fig. 13 shows the basic circuit.

Suppose point 2 is negative with respect to point 1. Electrons will flow from point 2 into condenser C_1 , forcing other electrons out; these go to point 3, then through VT_1 to point 1. (Of course, there is no path for these electrons through VT_2 .) This charges C_1 almost to the peak line voltage, with the polarity shown. The electron path is shown by the outline arrows.

On the next half cycle, point 1 is negative with respect to point 2. Now electrons cannot travel through tube VT_1 ; instead, they move from point 1 through condenser C_2 , tube VT_2 and



FIG. 13. The half-wave voltage doubler.

condenser C_1 to point 2, as shown by the black arrows. (This time, the path through VT_1 is blocked.)

Now just how does the voltage double? On the last half cycle, remember that point 1 was negative and point 2 positive. Since point 2 connects to the negative terminal of condenser C_1 , the voltage stored in C_1 adds to the line voltage, so that the voltage between points 1 and 3 is approximately double the line voltage. It is not exactly twice the line voltage, because C_1 is being slowly discharged by the load on this half cycle, but it is not far below double voltage if the load has a high resistance. This voltage is applied to tube VT_2 , so condenser C_2 is charged to almost twice the line voltage (and, of course, almost twice the line voltage appears between B+and B_{-}).

Since the current passes through rectifier tube VT_2 on only every other half cycle, this is a half-wave voltage doubler as far as condenser C_2 is concerned. Because of the half-wave action, the output voltage is somewhat less and the regulation is worse than a full-wave doubler.

▶ Notice that the B— terminal connects to one side of the power line in Fig. 13. This means that when the filament string is connected across the power line, there will be no d.c. volt-

age difference between the first tube cathode and its filament, and therefore no leakage current. We have thus improved upon the full-wave voltage doubler in one way, but have had to sacrifice some regulation to do it.

A complete half-wave voltage-doubler circuit is shown in Fig. 14. The action is identical to that of Fig. 13 as far as the B supply is concerned. Protective resistors R_1 and R_2 are used in series with the plates; also the pilot lamp P and its shunting section of the filament are between condenser C_1 and terminal 2—which is another protective feature. The current flow



FIG. 14. A typical commercial half-wave voltage doubler.

for the B supply helps to keep the pilot lamp lighted.

The filament circuit includes a resistor R_s to drop the voltage to the required amount for the tube filaments. The pilot lamp could be placed across a tapped section of R_s , but this is unnecessary because of the pilot lamp tap on the rectifier filament.

Voltage Multipliers

You have seen how the line voltage may be doubled by using charged condensers and rectifier tubes. This method need not stop here—we could have voltage triplers, voltage quadruplers, etc. Of course, every time the voltage is increased another step, the regulation becomes worse and the available current becomes less. For this reason, voltage doublers are the practical limit for most radio purposes. However, Fig. 15 shows how a voltage tripler can be made by adding an ordinary half-wave rectifier to a voltage doubler.

When terminal 1 is positive with respect to terminal 2, there will be an electron path from terminal 2 through condenser C_1 and tube VT_1 . At the same time there will be another path through tube VT_3 and C_3 back to terminal 1. Thus condensers C_1 and C_3 are charged on this half cycle.

On the next half cycle, electrons will move from terminal 1 into C_2 , forcing other electrons through VT_2 and C_1 . As in any half-wave voltage doubler, C_2 is charged on this half cycle to double the line voltage. This voltage is then added to that of C_s , so that three times the line voltage exists across C_2 - C_3 .



FIG. 15. A basic voltage tripler.

▶ A very important fact is that only a.c. sources can be used with voltage doublers, triplers, etc. No output can be obtained if a d.c. power line is connected to any of these voltage multipliers, because the polarity of a d.c. power line does not reverse.

Vibrator Power Supplies

So far, we have discussed power supplies which depend on a.c. or d.c. power lines. Power lines are not always available, however, so we must also consider other sources of power. Batteries are one such source, and can be used for portable or mobile equipment and for farm receivers, as will be discussed later.

In many locations, certain batteries are already available. On mobile devices such as automobiles, trucks, boats and airplanes, storage batteries of 6, 12, or 24 volts are used to operate the ignition and starting equipment; in many farm homes, 32-volt batteries are used for lighting purposes. These batteries furnish d.c. voltages which are adequate for filament supply but are not high enough to be used efficiently as plate and screen grid voltage supplies. We could use additional batteries, as was done in early installations, but means have now been discovered for raising the voltages of these batteries to a higher and more efficient value. Let us see how this can be done.

► Anyone who has not studied the fundamentals of radio might say, "use a step-up transformer," but you know that a battery delivers only d.c. Since steady d.c. provides no changes in flux linkage, there will be no voltage induced in the secondary if a battery is connected to the primary of a transformer. However, if d.c. flows through the primary in pulses, there will be changes in flux linkage and an induced secondary voltage will result. You will recall from your study of coils that when d.c. is first applied, the current rapidly builds up from zero to its maximum value and then flows at a steady rate. The flux linkages change as long as the current changes. Let's see how this idea can be applied to a battery and a transformer.

Suppose we connect a storage battery to the primary of a transformer as in Fig. 16A. When the circuit is first closed, current will flow through the primary: the resultant change in flux linkage will induce a large voltage across the secondary, as shown by 1-2 in Fig. 16B. The primary current will rise rapidly until it reaches its maximum value (determined by the resistance of the wire with which the transformer primary is wound), then remain constant until the circuit is broken. When the primary current becomes constant, the secondary voltage will drop to zero (2-3) and stay there, since voltage is induced only when there is a change in flux linkage.* When the primary circuit is opened, the current will cease flowing and the magnetic field will quickly collapse. This will cause another change in flux linkage, this time in the opposite direction, and a secondary voltage pulse will appear, as shown by 4-5-6 in Fig. 16B. This voltage will also drop to zero when changes in flux linkage cease.

▶ By using a step-up transformer and repeating this opening and closing regularly, we can get a series of alternate pulses from the secondary with high peak voltages.

From this, you can see that d.c. voltage can be stepped up with a transformer and changed into a.c. voltage if the primary circuit of the transformer is opened and closed fast enough with a switch. This allows the d.c. to flow in pulses which rise from zero to maximum and then decrease to zero again. Of course, if we could make the d.c. *reverse* in the primary, we would get more a.c. secondary pulsations, closer together and thus obtain a greater a.c. output average.

Full-Wave Vibrators

Fig. 17A shows how we can make the d.c. reverse. Notice that the primary is center-tapped. One battery terminal connects to this tap; the other battery terminal connects to a switch arm which connects first to one outside primary lead and then to the other.*



FIG. 16. A. C. can be produced by regularly interrupting a d. c. circuit.

Suppose the switch is thrown from position X to position Y. When the contact at X is broken, the current through P_1 drops to zero, as shown by line 1-2 in Fig. 17B. During the time it takes the switch arm to travel to position Y, the primary current remains zero. Then, as the switch arm makes contact at Y, we have current flow 3-4 through P_2 .

Since current flows in opposite directions through the windings P_1 and P_2 , a decreasing current in one produces a flux in the same direction as an *increasing* current in the other; therefore, the flux linkage change produced by current 1-2 decreasing through P_1 is in the same direction as the flux linkage change produced by

^{*} The primary current would probably be great enough to burn out the primary winding, if allowed to flow long.

^{*} One battery lead is grounded; it does not matter which one. The ground is shown because one terminal of a storage battery usually connects to the frame of the car or other device with which it is used.

current 3-4 increasing through P_2 . The change 1-2 produces a secondary voltage pulse like a-b-c in Fig. 17C. Then the change 3-4 produces pulse d-e-f in the same direction.

Between 4-5 in Fig. 17B, the current through P_2 is constant and there is no change in flux linkage. When the switch arm is thrown from Y to X, the current through P_2 drops to zero, as shown at 5-6, and when contact at X is made, current P_1 rises as shown by 7-8.

The changing flux due to currents 5-6 and 7-8 produces voltage pulses g-h-i and j-k-l. Then currents 9-10 and 11-12 produce pulses m-n-o and p-q-r. Thus, we now get double pulses in the secondary instead of the single ones of Fig. 16.

▶ We could apply the secondary voltage shown in Fig. 17C to a rectifier tube, pass the tube output through a filter, and get a fair d.c. voltage output. However, there are several things wrong with these voltage pulses.

One is that the voltage consists of narrow double pulses in each direction, while we'd rather have pulses which are broader and not as high. We can get them by speeding up the switch shown in Fig. 17A, and by moving its contacts closer together, so that very little time elapses while the switch moves from one contact to the other. This means that the flat parts of the curve in Fig. 17B, where the current is zero (2-3, 6-7, etc.) are practically eliminated, so the primary current change will be almost a straight line between points 1 and 4, 5 and 8, 9 and 12, etc. As a result, each pair of secondary pulses will merge together, and there will not be much of a gap between opposite pairs of secondary pulses.

Even so, these pulses are very high, sharp peaks. This makes the filtering job difficult. It also produces sparking at the switch contacts, because each rapid change of flux in the transformer induces a back-electromotive force (back e.m.f.) in the primary which tends to cause arcing between the contact and the switch arm. This will burn and soon ruin the switch contacts.

The Buffer Condenser. The sparking problem is easily solved by connecting a low-capacity condenser (called a buffer condenser) between the secondary terminals.* Now, when

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FIG. 17. Double voltage peaks are produced by using a full-wave interrupter.

the secondary voltage starts to rise, the condenser charges. This tends to reduce the sharpness of the peak in the secondary circuit, thus cutting down the back e.m.f. and thereby reducing the sparking at the vibrator contacts. Further, the discharge of the condenser when the secondary voltage begins to reverse helps to give us a secondary voltage more like that shown in Fig. 18—which has a shape approaching that which would be fed to a standard rectifier circuit.

Practical Vibrators

Fig. 19 shows how modern vibrators operate. The switch arm is a

^{*} The condenser is chosen to work with a particular transformer and a certain rate of switch closure. Its capacity is quite important. When replacing a defective condenser, use the capacity and working voltage originally used by the manufacturer.

thin, flexible, spring-like metal reed. As it vibrates, it alternately closes contacts N and O, thus alternately closing the circuit through halves of the primary.

The shaded area shows the "motor" used to drive the reed. This motor is



FIG. 18. The output of a vibrator-transformerbuffer condenser combination, which can now be rectified.

somewhat like the buzzer in an electric doorbell. Two types are used the separate driver type and the shunt type.

Separate Driver. The separate driver type is shown in Fig. 19A. The vibrating reed is between K and R. Contact M is a spring contact which touches the reed when the reed is at rest. When the ON-OFF switch S is closed, electrons flow through contact M and electromagnet L. This energizes the electromagnet, which pulls up the soft iron bar K fastened to the end of the reed. This moves the reed, opening contact M, which breaks the circuit through the electromagnet. Since the electromagnet is no longer energized, the reed moves back to its original position, again closing M. Current again flows through L, and the cycle repeats as long as the battery is connected. The period of vibration of the reed depends on its mechanical characteristics-its length, springiness, etc.

In moving back and forth, the reed closes contacts N and O alternately. Each closure completes the circuit through one of the primary winding sections. The buffer condenser C_1 in

the secondary prevents arcing at contacts N and O. This does not affect the motor circuit and so does not protect M, therefore a separate condenser C is used across this contact.

The contacts are large and flat-surfaced, made from a hard grade of tungsten to reduce pitting and burning to a minimum.

The Shunt Motor. The shunt motor shown in Fig. 19B does not require a separate contact and is more widely used than the separate driver motor. When switch S is closed, electrons flow through the chassis (shown by ground symbols), electromagnet L, and primary winding P_1 . The electromagnet then pulls the reed upward. closing contact N and connecting the battery directly across primary P_1 . Contact N shorts the coil L, and the current now flows through N and the reed instead of through coil L, so the releases: the reed electromagnet springs back. Its springiness carries it beyond the resting position-this overshooting causes it to close contact



FIG. 19. Separate driver and shunt vibrator motor units. One or the other of these will be found on practically all commercial vibrators.

O. Then current flows through P_2 . pacity When contact N is broken the current again starts to flow through L, and soon builds up enough magnetic pull to jerk the reed back and close contact N. This cycle of action is repeated as long as S is closed. We al----lines. ways have current flowing through P_1 (it flows through P_1 and L when N is open), but this current is much smaller than that when N is closed and does not have any appreciable effect on the wave form of the secondary voltage. $A = \frac{1}{2}$

Rectification

Now that we know how a high a.c. voltage may be obtained from a d.c.

pacity condensers in the filament and B+ supply leads, prevent this interference (called "hash") from being fed to the other tube circuits. Direct radiation is prevented by complete shielding, as shown by the dotted lines.

▶ Notice that the two battery leads are marked A-hot and A-ground. The one marked A-ground connects to the grounded side of the battery. In power packs using tube rectifiers, it makes no difference whether A+ or A- is grounded.

 \blacktriangleright If the filament current drawn by the rectifier tube can be eliminated the drain on the battery can be re-



FIG. 20. A complete vibrator power supply using a tube rectifier.

source, let us see how this a.c. voltage is rectified. Fig. 20 shows a typical full-wave rectifier circuit commonly used for this purpose. The only real difference between this and an ordinary power pack is that the filament of the rectifier tube is supplied from a battery. As one battery terminal is grounded, the filament is at the same potential as B—. This means good insulation is needed between the cathode and heater of the rectifier tube, for as much as 400 volts may exist between them.

Also, the voltage pulses which result from rectification are still rather sharp and the rectifying process generates r.f. noise signals. R.F. filters, consisting of r.f. coils and small caduced. For this reason, cold-cathode rectifier tubes are sometimes used in auto radios. To refresh your memory of these tubes, refer to Fig. 21, which shows the structure of a typical full-wave cold-cathode rectifier. The tube is filled with gas which, on ionization, becomes an excellent conductor.

Rectification is caused by the physical structure of the tube. When one of the pointed plates is positive it can pull many electrons out of the large cathode. But when the cathode is positive with respect to a plate, the cathode can pull only a few electrons from the small surface of the plate; as a consequence, only a comparatively small current flows in the undesired direction. In other words. the cold-cathode tube is a two-way conductor that conducts far better in one direction than the other.

The ionization of the gas in the tube causes a brilliant purplish glow which continually flickers as current through the tube changes. The coldcathode tube has a serious disadvantage in that it generates a large amount of r.f. noise signal-much more than the mercury-vapor tube which it resembles. Often this interference is difficult to eliminate because of poor joints at shield contacts to the chassis, changes in the tube characteristics, etc. If you meet this difficulty, the easiest way to solve it is to rewire the circuit for a heater type rectifier. Usually it is only necessary to wire in the filament circuit, retaining the original tube socket.

Synchronous Vibrators. Sometimes it is desirable to eliminate the rectifier tube altogether, particularly where space limitations are important. Let us see how we can do so.

Basically, a rectifier tube is just an electronic switch that reverses a circuit connection every time the a.c. cycle reverses, and so keeps current flowing always in the same direction through the circuit. We can replace this electronic switch with a mechanical one—a vibrator—and get the same effect. Of course, this vibrator must be synchronized with the vibrator in the primary circuit so that both of them open and close their circuits at the same time. We can get this synchronization by mounting both sets of vibrator contacts on the same vibrating reed. This also helps to save space, by making two reeds unnecessary.

Fig. 22A shows the fundamental operation of such a circuit. Switch SWis synchronized with the vibrator, and replaces the rectifier tube. Notice that more is involved than a simple replacement of the tube by the switch; the circuit is also changed around, with the rectifying element in the negative side of the output circuit. This is necessary because one side of the battery is grounded, which means that the vibrating reed is also grounded. As the same reed is used in SW, if we connected the switch in the positive side of the output (as we do with a tube) we would be grounding B+.

When current flows through the primary in the direction shown by the



FIG. 21. The elements of a gaseous or coldcathode rectifier tube.

solid arrow, the polarity of the secondary voltage is that shown by the solid arrows drawn between the secondary terminals. Current flow through the other half of the primary reverses the secondary voltage polarity, as shown by the dotted arrows.

Notice that the center tap c on the secondary will be positive with respect to a when current flows one way through the primary, and positive with respect to b when current flows the other way. Obviously, switch SWmust connect a to ground when a is negative with respect to c, and b to ground when b is negative with respect to c, in order to maintain the right polarity for the load $R_{\rm L}$. Therefore, it must be synchronized with the primary vibrator. This is the reason these types are called "synchronous" vibrators, while the other type is known as the "non-synchronous" type.

Fig. 22B shows the complete circuit of a practical synchronous vibrator, with both switch units operating from a common vibrator reed.

Either a separate driver or a shunt motor unit may be used, with the shunt type predominating. Notice that the whole unit is enclosed in a shield (dotted lines) and that r.f. filters L-C are used in both the B+ and filament leads.

▶ You must be careful about the battery connections when you use one

This reverses the B+ and B- terminal polarities, which prevents operation of the radio device. Further, this reversed polarity will ruin the electrolytic condensers if it exists for more than a few seconds.

Different polarities exist in various cars, boats, etc., where the batteries are used for ignition and other pur-



FIG. 22. The elements of a synchronous vibrator which uses mechanical rectification.

of these synchronous vibrators. When the battery polarity is reversed, the direction of current flow in the primary windings reverses, which interchanges the secondary polarities. Battery connections do not matter in a vibrator power pack which uses a rectifying tube—the tube automatically passes current in the proper direction, as whichever plate is made positive will pass current.

The synchronous vibrator pack is put together with a particular polarity in mind. Should the battery polarity be reversed, the secondary contacts would be closing to the *positive* terminals (with respect to c in Fig. **22**A) instead of the negative terminals. poses. As the battery cannot be reversed, some provision is normally made for correcting this trouble, usually by providing a means of reversing the secondary contacts. A terminal board connection, or a vibrator which can be reversed in its socket is commonly used. Always check the vibrator output after the battery has been connected. If the output has the wrong polarity, reverse the connections by the means provided.

▶ While vibrator systems are most often found in cars, small boats and planes, they are also used in some farm receivers and may operate from a 2- to 6-volt storage battery or from a 32-volt farm lighting plant. In the latter case, the receiver tube filaments are generally wired in series, as in an a.c.-d.c. receiver, to operate directly from the 32-volt supply.

Modern Battery Receivers

When you get into servicing, you'll find that receivers using batteries for their power supplies are very common; in fact, because of the wide popularity of modern portable radios, there are many more battery-operated sets in use today than there were in the days when batteries were the only source of power available. All these sets use A, B and C batteries—or suitable substitutes—to supply filament, plate and grid operating voltages. Let's look into the various ways these batteries may be used.

Filament Voltages

The cost and trouble of replacing worn-out batteries has always been one of the main objections to batteryoperated receivers. To combat this objection, engineers developed battery tubes which required lower and lower current drains. The first battery tubes, for example, required 5 volts to supply their filaments. Since each tube drew .25 ampere (250 ma.), each filament consumed 1.25 watts. Then 3.3-volt tubes* which drew .132 ampere (132 ma.) made their appearance, followed by 2-volt, .06-ampere (60 ma.) types. Modern tubes need 1.4 volts and draw only .05 ampere (50 ma.)—a power consumption of only .07 watts, and the future will undoubtedly bring forth further progress in this line.

The filaments are generally connected in parallel to an A battery which furnishes about the correct voltage. If this voltage is too high, a series resistor is used to cut it down. Fig. 23 shows such a circuit; resistor R is used to drop the voltage to the correct amount. The difference in voltage between that required by the tubes and that furnished by the battery is the voltage which the



FIG. 23. Battery tube filaments are frequently connected in parallel.

resistor must drop. As the tubes are in parallel, R must carry a current equal to the sum of the filament currents. The value of R is then equal to the required voltage drop divided by the total current.

Sometimes tube filaments requiring the same current are wired in series and supplied by an A battery which will deliver the sum of the tube filament voltages. This connection increases the A voltage required, but reduces the current drain to that used by only one tube, so the wattage (voltage \times current) furnished by the battery is no greater than for parallel filament operation—and the lower current drain lengthens the life of the battery.

^{*} It is common practice to speak of tube types by their filament voltage rating. Thus, we may have 50-volt tubes, 1.4-volt tubes, 6.3-volt tubes, etc.

Grid Bias Voltages

Usually the control grid of a tube requires a bias to make the tube operate properly. This bias voltage normally makes the grid negative with respect to the cathode (or the filament in battery tubes).

The C Battery. A typical C battery arrangement for furnishing bias is shown in Fig. 24. This battery is made up of a series-connected group of 1.5-volt dry cells. It may have only two terminals, or it may have a number of taps to supply tubes needing different C bias voltages. Notice that C+ connects to A-, so the bias is applied between the grids and the negative sides of their filaments. The grids of VT_1 and VT_2 are connected together and have a bias voltage of -1.5 volts; the grid of tube VT_3 is biased 3 volts negative, while -4.5 volts is applied to the grid of tube VT_4 .

Since the grid circuit does not ordinarily draw current, the C battery usually has a long life.

The Bias Cell. Various means have been worked out to eliminate the C battery and thus save space in a receiver. A cell known as a "bias cell" is frequently used in battery sets —and also in a.c. receivers where small amounts of bias are needed and hum may develop if other supplies are used.

Essentially, the Mallory grid bias cell is a very small electrochemical cell with a potential of approximately 1 volt. Its current-supplying capacity is less than 1 microampere, but it can furnish voltage indefinitely as long as no current is taken from it. This makes it suitable for the grid circuit, where normally no d.c. is drawn from the bias supply. If strong signals drive the grid positive, the resulting grid current flow will be in such a direction as to recharge the bias cell and thus help prolong its life.

► The cell consists essentially of an acorn-shaped metal cap or shell filled with an electrolyte, and a black disc which seals the circular end of the shell. The shell is the negative electrode, the black disc the positive electrode. The disc is insulated from the shell and sealed in place by a rubber gasket around its edges.

Suitable mounting brackets have been developed for this cell, usually consisting of a cup in which the cell fits and a spring contact which presses against the black electrode and holds the cell firmly in position. Mounting brackets holding as many as nine cells in series are available for applications requiring more than 1 volt. The cell is connected just like the C battery, except that its small size makes it possible to mount it right in the receiver.

▶ Bias cells must be checked with a vacuum tube voltmeter which draws no current, since the cell can't supply enough current to operate an ordinary type voltmeter. When replacing a bias cell, slip the old cell out of the mounting and slip in a new one Make sure the spring arm has enough tension to make firm contact with the black electrode.

Convection Current Bias. Another method of biasing a tube depends on the fact that some of the electrons flowing in a tube hit the control grid wires and are collected by the grid. These electrons flow out of the control grid lead, through the external circuit, and back to the cathode. This flow is called a "convection current." If a resistor is placed in the control grid circuit, the electrons flowing through it produce a voltage drop which biases the tube. The end of the resistor connected to the grid is negative, as the electrons enter this end.

The convection current varies with plate current, increasing with increases in plate current. Even at best, however, this current is so small that it is measured in microamperes, so a large resistor of 10 or 15 megohms is necessary to get enough bias.

Like the bias cell, the convection current method of biasing may be used with any type power supply. It is ordinarily used in a.f. amplifier stages.

Automatic C Bias. Another method of biasing the tube without the use of a C battery is shown in Fig. 25. Electrons flow from each tube filament to its plate, then to B+ and through the B supply to B-. In order to get back to the filaments,



FIG. 24. Grid bias paths from a C battery.

the electrons must go through resistor R to the chassis, then through the chassis to the grounded terminal of the filament circuit. Therefore, the plate current from all of the tubes flows through resistor R. This produces a voltage drop across the resistor with the polarity shown. (Remember, the end of the resistor which electrons enter is always negative.) Suppose we bring the grid return of tube VT_{4} to the negative side of resistor R. Then, the drop across the resistor makes the grid negative with respect to its filament-in other words. biases the tube. We can connect any number of grid returns to resistor R.

Of course, the bias voltage we get depends on the amount of plate current flowing and the value of the resistor in ohms.

This method is called "automatic bias," because the bias will auto-



FIG. 25. The voltage drop produced by plate current flow through R produces an automatic C bias.

matically adjust itself for plate current changes, increasing when plate current increases and decreasing when plate current decreases. Notice that this method makes the B supply furnish the bias voltage.

Filament Biasing. One of the important reasons for connecting tube filaments in series is the fact that it is possible to get a bias voltage from the filament voltage drops. Fig. 26 shows how this can be accomplished. Here, the plate and grid signal circuit parts have been omitted for simplicity.

This circuit contains four 1.4-volt tubes, with the filaments in series. A 6-volt A pack is used, so there is a voltage drop of 1.5 volts across each filament, and .05 ampere flows through the complete circuit.

Tube VT_1 has its grid returned to point 1, which is the negative side of its filament. There is no bias on this tube, as there is no voltage difference between the grid and filament:

There is a drop of 1.5 volts between points 1 and 2. Since point 3 is connected to point 2, this same drop exists between 1 and 3. Therefore, if we connect the grid of VT_2 to point 1 there will be a bias voltage of 1.5 volts between this grid and point 3 on the filament of VT_2 . Similarly, the 1.5volt drop between points 3 and 4 can be used to bias tube VT_3 .



FIG. 26. The d. c. voltage drop across seriesconnected battery tube filaments can be used for bias too.

The grid of tube VT_4 is connected back to point 1. There is a 1.5-volt drop between 1 and 2, a similar drop between 3 and 4, and a third 1.5-volt drop between terminals 5 and 6, making a total of 4.5 volts between point 1 and point 7 (the filament of VT_4). Therefore, tube VT_4 has a bias of 4.5 volts.

The **B** Circuit

The B supply circuits in a modern battery receiver are very simple. The B+ supply leads are joined together and are connected to the B+ battery lead. The B- terminal usually connects to the A- filament terminal. The screen grid and other positive electrodes may connect to B+ or to a tap on this battery. Thus, the one battery supplies all the needed positive voltages. A series resistor may be used instead of a battery tap to provide screen grid voltages, particularly on the more modern sets. A bleeder is never used in battery sets. because the bleeder current would shorten the life of the B battery.

Typical Battery Sets

Now let's put together the things you've just learned, and study the complete battery supply circuit in Fig. 27.

Two switches are "ganged" (connected) together in most battery sets and are used to turn the set ON or OFF. In Fig. 27, switch SW is the actual ON-OFF switch. When turned OFF, this switch opens the filament circuit and thus stops tube emission, which prevents the flow of plate current. The second switch, SW_1 in Fig. 27, serves to open the B supply circuit to prevent leakage paths from draining the B battery. This is necessary because, in many sets, by-pass condenser C_1 is an electrolytic, used to keep signal currents out of the B battery. Since all electrolytic condensers have a certain amount of leakage, there will be a constant drain on the B battery un-



FIG. 27. Three biasing methods are shown here: 1, bias cell; 2, convection current; and 3, automatic bias.

less the B circuit through C_1 is opened by switch SW_1 .

With the switches closed, current flows through the tube filaments, heating them to the point where they can emit electrons. Plate current then flows from B— through switch SW_1 , resistor R, and the chassis to the grounded tube filaments. The electrons return to B+ from the various screen and plate circuits.

In passing through resistor R, the plate and screen currents set up a voltage drop with the polarity marked



FIG. 28. Supply circuits in a typical modern battery portable receiver

on condenser C_2 . The positive end of this resistor connects to the filament of the 1A5 through ground and its negative end connects to the 1A5 control grid. The voltage across R thus biases the 1A5 control grid.

Resistor R_1 has a value of 10 megohms. Convection current through it produces a bias voltage for the 1H5 control grid. A bias cell provides C bias for the two 1N5 tubes. Ordinarily, all three methods would not be used in the same chassis, but we have grouped them together here for convenience in illustration.

A Series Filament Circuit. The circuit shown in Fig. 28 appears complicated because the tubes are not arranged in the same order as their filament connections. The circuit may be more easily understood by referring to Fig. 29, which shows just the filament circuit and its associated parts.

You can readily see that these filaments carry more current than is supplied them by the A battery. Since the filaments are also the cathodes of the tubes and are connected in series, each one carries the plate and screen currents of all the tubes connected after it. Our circuit has a 1T5 power output tube which has a fairly high plate and screen current. In order to prevent this current from overloading the filaments of the other tubes, we use resistor R_1 to shunt part of the plate current of the 1T5 around the filaments of the 1A7, 1N5 and 1H5.

Tracing the current flow in the circuit, starting from A— and B—, you see that the filament of the 1H5 carries its regular current plus the plate currents of tubes 1A7 and 1N5, plus part of the plate current of the 1T5 (the rest goes through R_1). The 1A7 filament carries its own current plus the plate current of the 1N5 and part



FIG. 29. The filament circuit of Fig. 28 shown in simplified form.

of the 1T5 plate current, and the 1N5 filament carries both its own current and the part of the 1T5 plate current which did not go through R_1 . All these plate currents are the normal steady currents, which can usually flow through other parts of the circuit without causing any harm. But the signal currents (which, as you know, are variations impressed on the steady plate current) can't be allowed to flow through the filaments of tubes other than their own, for this would cause undesirable coupling between stages.

Therefore, we provide condenser paths for the signal currents between the tube filaments and B—. The size of each condenser must be chosen so that the impedance it offers to the signal current it is supposed to carry will be far less than the resistance offered by any path through the filaments. Since B— is grounded to the chassis, we ground one end of each condenser to the chassis and connect the other end to the filament of the tube with which it is to be used. Thus, the signal current of tube 1A7 passes through condenser C, that of 1N5through C_1 , and that of 1T5 through C_2 . Since 1T5 is a power output tube and carries an audio frequency plate current, we must use a high-capacity electrolytic condenser for C_2 . It should be connected with the polarity shown. No condenser path is provided for tube 1H5, because its signal current flows directly to it without having to go through any other tube first. Referring back to Fig. 28, you see that the 1T5 is biased by grounding its control grid return, thus connecting it to A— (which is 4.5 volts negative with respect to the 1T5 filament). No external bias source is used for the 1H5 and 1N5 tubes. The 1A7 control grid (fourth grid from the filament) is connected through resistor Rto the filament of the 1H5, which connects to the filament of the 1A7.

You can readily see that any voltage drop across resistor R will make the 1A7 control grid negative with respect to its filament. Now, when a signal (an a.c. voltage) is applied between diode plate D of the 1H5 and its filament, the tube acts like an ordinary rectifier for which resistor R is the load. The amount of rectified voltage across R depends on the

strength of the signal. The greater the signal voltage, the greater the voltage across R and the greater the bias on the 1A7. Since this bias controls the gain of the 1A7 tube, we have a circuit in which the gain goes down for strong signals and goes up for weak ones—in effect, an automatic sensitivity control or, as radio men call it, an automatic volume control (a.v.c.). You'll study the automatic volume control in great detail later on, but even now you can see the principle on which it works.

The Three-Way Receiver

The chief use today for batteries is in portable receivers. To save batteries, it is desirable to be able to operate a portable on a power line whenever the set is used where a line is available. In other words, we'd like to have a three-way power supply which can use a.c., d.c. or batteries. Let's see what kind of a circuit we need to get it.

In the receiver shown in Fig. 28, we need 6 volts for the filaments and 90 volts as the B supply. The latter can easily be obtained from a power line by using the familiar a.c.-d.c. half-wave rectifier supply shown in Fig. 30. Between the B+ and Bterminals there will be approximately 90 volts, so these terminals may be connected directly to the B+ and Breceiver leads.

▶ The lowered filament current demand of modern tubes makes it possible for this rectifier also to supply power for the filaments, something not possible in the early days of radio. Modern tubes only require .05 amp. (50 ma.) for filament operation, so when in series, this is a value well within the rectifier tube limits.

A voltage divider consisting of R_2 and R_3 is used across the rectifier terminals to provide the filament voltage. The 6 volts needed for the A supply appear across resistor R_2 . R_2 acts as a bleeder, making the A supply voltage more or less independent of tube drain.

The rectifier output is filtered by its own filter and also by the filament by-pass condenser C_2 in Fig. 29. This



FIG. 30. This universal power supply can be used in battery sets to form a battery—a. c.-d. c. receiver.

condenser acts as a filter for tubes 1A7, 1N5 and 1H5, because it is across the A supply as far as they are concerned. It does not act as a filter for the 1T5, however, because the 1T5 comes between the A supply and the points where C_2 bridges across the A supply circuit. Therefore, some hum currents flow through the 1T5 filament; since this is the output tube, there is little amplification of the hum before it reaches the loudspeaker, so a small amount of hum is permissible.

▶ There is an important precaution you should remember when servicing three-way receivers. Never pull out and reinsert a tube in its socket while the receiver is turned on.

In receivers which have a bleeder resistor across the A supply (like R_2 in Fig. 30), doing so will usually cause no trouble. But suppose you pull

out a tube in a set having no bleeder across the A supply-what will happen? This breaks the circuit so there will be no filament current through the series resistor R_s , and the voltage between the A+ and A- terminals rises to practically the B supply voltage. If the pulled tube is any except the 1T5, the condenser C_2 will charge up to this high voltage. When the tube is reinserted, this condenser discharges through the filaments; since C_2 is a large condenser, the discharge current may be great enough to burn out one of the filaments. To be on the safe side, turn the receiver off while removing and replacing tubes.

▶ We will not concern ourselves here with the various switching arrangements used to disconnect the batteries when the set is operated on a.c. or d.c., as these will be covered later. As a matter of fact, the batteries are sometimes left in the circuits at all times, and the output of the power pack is connected right across them. (No bleeders are used in such circuits, since they would drain the batteries. However, the batteries themselves act as bleeders across the power supply.)

This prolongs the life of the batteries, for the power pack sends a reverse current through the batteries at the same time it supplies A, B and C voltages to the tubes. Of course, the current through these dry batteries cannot recharge them, but it does drive off the hydrogen bubbles which form around the carbon electrodes. This reduces the internal battery resistance and helps the batteries last until their active ingredients are used up.

Special Power Supplies

Sometimes neither power lines nor batteries can supply the power we need. For example, the power line may have the wrong voltage or frequency for the equipment we want to operate, and the device may need far more power than can be readily supplied by batteries. Let's see what special power supplies have been developed to meet such a situation.

Engine-Driven Generators

One way to get the power we want is to make it ourselves with a gasolinedriven generator. Usually we choose to develop 110-volt, 60 c.p.s.* power, so standard, easily-obtainable equipment may be employed. Most a.c. equipment is built for this frequency and voltage because they are found throughout the U.S.A., except for the relatively few districts having d.c., 25cycle or 40-cycle a.c. lines.

► A typical gasoline-engine-driven a.c. generator is shown in Fig. 31. This type is used by servicemen in mobile public address installations, which usually require fairly large amounts of power. The unit is entirely self-contained and is furnished with a gasoline storage tank. Various sizes can be obtained, but a 300-watt unit is quite common and satisfies most needs. A 6-volt battery (usually a storage battery) is the only auxiliary equipment. It is needed to excite the electromagnets of the generator and to operate the ignition system of the gasoline engine. In an automobile installation, the car battery may be used. The output of the generator has a simple spark filter

(coil and condenser) to suppress interference.

Operating Equipment on Supplies for which They Were Not Designed

In your service work, particularly if the power line in your community is not the standard 60-cycle, 110-volt variety, you may be asked to make a set run on power for which it was not designed. Let's take up some of the most common problems of this sort, and see how you can solve them.

Adapting to a Line of a Different Frequency. Most a.c. power lines have frequencies of 25, 40 or 60 c.p.s. Occasionally you will have to install a radio receiver using a power transformer on a line having a frequency other than that for which the receiver was designed. Can a receiver de-



FIG. 31. A gasoline engine-driven generator of the type used in radio and public address work.

signed for one frequency be used on another? Yes and no!

For one thing, a receiver designed for 60 c.p.s. operation will not usually have a ripple filter sufficient to handle 40 to 25 c.p.s. Even more important, transformer equipment designed for low frequencies will work at higher frequencies (within reasonable limits) but the reverse is not true. That is, a 25 or 40 c.p.s. receiver will work with

^{*} The abbreviation "c.p.s." stands for "cycles per second." Radio men frequently shorten this expression to "cycles," calling a 60 c.p.s. line a 60-cycle line.

a slight change on 60 c.p.s., but a 60 c.p.s. receiver should never be run on 25 or 40 c.p.s. lines. They may operate for a while, but in a short time the transformer will burn up.

The reason is that most transformer primaries are built to have a fairly high reactance at the frequency on which they're designed to operate; this prevents the primary from drawing large currents from the line. Lowfrequency transformers, for example, have large, heavy iron cores to help give them the reactance needed. Now as you have already learned, the reactance of a coil is less the lower the frequency of the voltage applied to it. Therefore, a primary designed for 60cycle lines will have a much lower reactance when it is used on 25- or 40cycle lines, and so will draw much more current than it is designed to This current overload will carry. overheat the transformer and eventually burn out its windings.

Therefore, to operate a 60 c.p.s. receiver from a 25 c.p.s. line, the power transformer must be replaced with a 25 c.p.s. transformer having the same output ratings. Then extra condensers must be added in parallel with the filter capacities to increase the amount of filtering. If the pack has tuned filters, they must be retuned to 25 c.p.s. or the tuning parts removed.

▶ When a 25 c.p.s. or 40 c.p.s. receiver or amplifier is to be operated from a 60 c.p.s. line, the transformer will be more efficient at the higher frequency (because of its large core) and therefore will deliver a higher voltage than normal. If voltages more than 10% above normal are produced, you should use a line regulator (variable resistor) in series with the primary, and adjust the resistor until the voltages are normal. Special resistors for this purpose are available from radio supply stores. **D.C. Equipment on A.C. Lines.** Often people move from a 110-volt d.c. district to a 110-volt a.c. district. In this case it is best to recommend getting an a.c. receiver, because it will be much better than the d.c. receiver. But if the customer is satisfied with his d.c. receiver (and it is only a d.c. set, not a universal receiver), you could recommend a small a.c. motor-driven d.c. generator. This



FIG. 32. Either the rotary converter at the left or the inverter at the right can be used to obtain 110-volt, 60-cycle a. c. power from d. c. power lines. The inverter is a high-power vibrator unit, using no rectifier.

equipment is usually more costly than a new receiver, however.

A.C. Equipment on D.C. Lines. As a rule, a.c. receivers are so much better than d.c., universal and battery receivers that it is quite common to adapt a good a.c. receiver to 32-, 110and 220-volt d.c. lines instead of buying a receiver designed for these special voltages. Two procedures are possible. Use a d.c. to a.c. rotary converter (a combination d.c. motor and a.c. generator) or a magnetic vibrator type d.c. to a.c. converter (often called an inverter). These devices are shown in Fig. 32.

For the average receiver, a 100watt converter or inverter is large enough. Adaptation is simple—just insert the receiver power plug in the receptacle of the converter, then connect the converter to the d.c. line. If a ground terminal is provided on the converter, connect a wire from it to the regular ground. The rotary converter is costly, but has a long life and can be had in any desired power rating for receivers or public address equipment; the magnetic vibrator inverter is comparatively inexpensive, but its power capacity is limited to 200 watts maximum, and the vibratormust be replaced about once a year.

Power Line Operation of Battery Sets

You may often wish to operate a battery receiver (not a three-way receiver) using 1.4-volt or 2-volt tubes from the a.c. power line. Power packs are available which do not take up any more room than the original bat-



FIG. 33. This compact A-B-C power unit can be used in place of batteries so power-line operation can be obtained.

teries; in fact, the batteries may be removed and the power packs substituted for them. A typical power pack is shown in Fig. 33. A full-wave rectifier tube is used with an untuned filter to provide B and C voltages. The B section of the voltage divider is tapped to provide any voltage normally required, while various values of C voltage are obtained by adjusting the slider on the lower section of the divider.

A copper-sulphide (similar in action to a copper-oxide) type rectifier is used in a full-wave bridge circuit for the A power. Filters in cascade furnish hum-free A voltage.

Copper-sulphide, copper-oxide and a recently developed selenium-iron unit are metal discs, specially prepared to act as rectifiers. These rectifiers have very low voltage drops and can be designed to carry high currents, so are better than receiver-type tubes where tow-voltage, high-current power is needed. Vacuum tube rectifiers have an internal drop of 15 or more volts, which may be half or more of the available voltage in a low-voltage circuit and so represents a large power loss. (Tubes are better in high-voltage circuits, where 15 volts may be only a 3 or 4% loss and where disc rectifiers cannot withstand the high peak inverse voltages.)

These units have the characteristic of letting electrons flow in one direction much easier than in the other. The elements are bolted or clamped together with insulating washers in such a manner that all four sections of a bridge make a single unit having four connections.

Lesson Questions

Be sure to number your Answer Sheet 13FR-2.

Place your Student Number on every Answer Sheet.

Send in your set of answers for this lesson immediately after you finish them, as instructed in the Study Schedule. This will give you the greatest possible benefit from our speedy personal grading service.

- 1. What is the maximum possible voltage which could be obtained from a universal a.c.-d.c. power supply, operated from an a.c. 115-volt power line, with no load?
- 2. Why is a condenser-input filter used instead of a choke-input filter, in a universal a.c.-d.c. receiver power supply?
- 3. What is the purpose of placing a resistor at point a in Fig. 3?
- 4. Suppose you are replacing the pilot lamp in a universal receiver. Is the current rating of the replacement lamp important?
- 5. Can a transformerless receiver using a voltage doubler be operated from a d.c. power line?
- 6. In connecting a receiver having a non-synchronous vibrator and tube rectifier, does the polarity of the storage battery matter? $U \neq S$
- 7. Name five methods of getting C bias in a battery receiver.
- 8. Why is switch SW_1 used to open the B supply circuit when turning off a battery receiver like the one shown in Fig. 27?
- 9. What type of meter must be used to check a grid bias cell?
- 10. Why are copper-oxide (or selenium) rectifiers always found in low-voltage high-current power supplies, instead of vacuum-tube rectifiers?

29 World Radio History

FIRST IMPRESSIONS

First impressions mean a lot in this busy world. An applicant for a job has a pretty tough time making the grade if his appearance and first few words do not make a favorable impression on the employment manager. A salesman likewise gets the "cold shoulder" if there is anything about him which annoys the prospect.

With technical material of any kind, however, first impressions can be very treacherous. Oftentimes a simple technical book will contain a number of apparently complicated diagrams, charts, graphs, sketches or tables. Since we glance mostly at illustrations when inspecting a book, we are apt to get a misleading impression. Also, paragraphs, pages or entire lessons may seem difficult during the first reading, but become almost magically clear during the second or third reading.

If first impressions of a required task are favorable, fine and dandy; if unfavorable, don't be discouraged, but wade right into the work and give it a chance to prove that first impressions don't always count.

J. E. SMITH.



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STUDY SCHEDULE NO. 14

For each study step, read the assigned pages first at your usual speed. Reread slowly one or more times. Finish with one quick reading to fix the important facts firmly in your mind, then answer the Lesson Questions for that step. Study each other step in this same way.

□ 1. Introduction to Acoustics
2. Technical Facts about Sound and Hearing
3. Sound Pickups and Reproducers
4. Requirements of an Audio Amplifier
5. The Fundamentals of Television
6. Video Amplifier Requirements
7. Mail your Answers for this Lesson to N.R.I. for Grading.
8. Radio Receiver Troubles—Their cause and Remedy
9. Start Studying the Next Lesson.

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HOW SOUNDS AND SCENES ARE CONVERTED INTO AND REPRODUCED FROM ELECTRICAL SIGNALS

Introduction to Acoustics

RADIO, as you know, is a means of sending intelligence through space. Since it requires no man-made paths between sending and receiving points. it is obviously more desirable in many cases than communication systems which send the same intelligence over land wires or cables. Although land wire telephone, telegraph, and picturetransmitting systems are not of particular importance to radio men, they differ fundamentally from radio systems only in the method used for transmitting the intelligence from the sending to the receiving point. In fact, research work carried out by telegraph and telephone companies has contributed much to the improvement of radio apparatus and technique.

▶ The three forms of intelligence which can be sent over either wire or radio communication systems are: 1, sound, such as that used in radio broadcasting, in radiotelephone communication, and in telephone systems; 2, pictures, either still or moving, such as those transmitted by radio or land wire television and facsimile systems; 3, code, such as the dots and dashes used in radio and wire telegraphy.

Code transmission can be obtained simply by opening and closing a key in the transmitter circuit which, in the case of radio, causes the carrier current to be fed intermittently to the antenna and which, in the case of wire telegraphy, sends pulses of current over the land wires.

On the other hand, sounds and scenes

represent intelligence which must be converted into equivalent electrical signals before being sent over wires or given "rides" through space on radio carriers. At the receiving end, the intelligence signals pass through many electrical circuits before being restored into the original sounds or scenes.

Thus, we must have a device capable of responding to all the characteristics of the sounds or the scenes, so that it can render a faithful electrical reproduction of the desired intelligence. Also, we must have reproducing equipment which can be fed an electrical signal and which will reproduce the original sounds or scenes as faithfully as is economically practical.

Furthermore, amplifiers must be used in both the transmitter and the receiver, in order to increase the voltage and the power of the electrical signal. Since the electrical signal characteristics, in turn, depend upon the sounds or the scenes, we must learn something about the characteristics of sound waves and light waves. First, we will study sound waves.

SOUND IS A WAVE MOTION

You know that sound can be produced by striking metal objects together, by rustling stiff paper, by vibrating your vocal chords, by plucking a stiff wire or spring, or by any one of thousands of different methods. Yet, though the ways we can create sound appear to be many and varied, they are basically identical. In every



FIG. 1. When a piano key is depressed, a "hammer" strikes a piano wire (actually a group of wires). The wire is set into vibration; this vibration is transferred to the surrounding air molecules. The alternate compressions and rarefactions of the air result in sound waves which travel out from the wire in all directions. (For clarity, only one direction is indicated here.)

case, we create sound only if we produce a vibration. The vibration may be of very short duration, as when two stones are struck together, or it may be prolonged, as when a bell is rung but, whether short or long, a vibration must exist before sound can be produced.

► However, a vibration alone is not enough to make a sound. The following is a favorite experiment in highschool physics classes: An alarm clock is suspended in a glass jar. The jar is then sealed except for a hose connection leading to an evacuation pump. Provided enough of the air in the jar is removed, no sound is heard when the alarm clock goes off—even though you can actually see the clock hammer strike the bell. This is an illustration of another basic fact about sound: no sound is produced by a vibration which occurs in a vacuum. There must be an elastic medium, capable of conducting the sound vibrations, in contact with the vibrating object before a sound is produced. This medium may be a gas, like air, or a liquid, like water, or a solid, like steel—in fact, it may be anything but a vacuum.

We know, then, that we must have some vibrating object and some medium in contact with the object before we can have a sound. Scientists tell us that the sound itself consists of a vibration of the medium. The vibrating object is the source of sound, the medium is the substance through which the sound is transmitted, but the sound itself is not a physical object like either of these; it is a vibratory motion.

► Let's make this clearer with an example. Suppose we consider what happens when a piano wire is struck by its hammer. Fig. 1A shows a piano wire, its hammer, and a schematic representation of a few of the air molecules which surround the wire. If we strike the piano key, the hammer hits the wire and bows it outward. The moving wire strikes the air molecules a in front of it and drives them forward; they then strike molecules b, which in turn are driven forward to strike molecules c, and so on. In this way, the motion of molecules a will be transferred all the way along the line of molecules. In fact, if it were not for the inevitable losses of energy which occur in such an operation, the forward motion of the molecules would be transferred indefinitely.

However, it is only the *motion* which is transferred in this manner; the individual molecules lose most of their energy when they strike the next molecules in line, and soon come to a stop. In other words, molecules do not travel from the source of sound to you—they pass along energy from one to another.

Fig. 1B shows the molecules as they would look an instant after the hammer has struck the wire. As you can see, the forward motion of the molecules has produced a compression of the air in the region of molecules a, b, and c, while farther on, where the motion has not yet reached, the molecules are still their normal distances apart (the air is uncompressed).

After it has bowed as far forward as it will go, the piano wire snaps back toward its original position. However,

Fig. 1C shows the distribution of the molecules when the piano wire has reached the end of its backward swing: molecules a, b, and c are now farther apart than normal, because they are still moving to fill up the vacuum created by the backward motion of the wire; hence, the air in this region is rarefied. (Rarefy means to make less dense; it is the opposite of compress.) Molecules d, e, and f have already gone through their forward motion. and are just starting on their backward motion. Therefore, they are about their normal distances apart, and the air in this region is neither compressed nor



Here is experimental proof that sound-producing bodies vibrate. The tuning fork will produce sounds when struck and forced into vibration. If a ball is suspended so that it touches the vibrating fork, it will bounce away, proving that the fork is in motion.

because of its elasticity, it does not just come back and stop; instead, it overtravels and bows backward. Its backward motion drives away the air molecules which were originally behind it, leaving a partial vacuum in its path. Molecules a, which by this time have stopped their forward motion, now rush back to fill up this vacuum; molecules b then come back to fill the space vacated by molecules a, molecules c come back to take up the space just vacated by molecules b, and so on. We now have a backward motion of the air molecules which, like the previous forward motion, will be transferred all along the line of molecules.

rarefied. Molecules g, h, and i are still going through their forward motions, so the air in their region is compressed. Thus, distant molecules are made to move back and forth in the same manner as those at the source of the sound.

The piano wire will snap back and forth at least several more times before it comes to rest, and each complete back-and-forth motion will produce two motions which will be transferred through the air molecules surrounding the wire—a *forward* motion (which we can call a compression wave) and a *backward* motion (a wave of rarefaction). Remember: the individual molecules move forward only until they strike other molecules, and move backward only until they are struck by the next wave of forwardmoving molecules.

► Let's pretend that the piano wire produces a pure tone having sine-wave. characteristics. (Actually it doesn't produce a pure sine wave, but let's see what could happen if this were true.) Then, Fig. 2 illustrates the movements of the molecules after the piano wire has vibrated several times. Instead of picturing just two lines of molecules in this figure, we have drawn curved vertical lines which represent millions of ticular instant. The horizontal line represents normal air pressure. At points of rarefaction, the air pressure is below normal; at points of compression, it is above normal. The final result, when we draw a curve through all the points constructed, is a simple sine wave.

► However, whether or not the sound wave is a sine wave, regardless of its form, the air molecules will be alternately compressed and rarefied. The plot of air pressure thus follows the vibrations of the source, so that the plot shows the exact wave shape. This



Another example of wave propagation. When ball I is struck by the cue at the left (in A) it moves forward, striking ball 2. Ball I then stops rolling as its energy has been imparted to ball 2, which now strikes ball 3, etc. As shown in B, C, and D, the balls move successively as energy is transferred from ball to ball. Sound energy is transmitted from molecule to molecule in much the same wav.

molecules; each molecule on a given line is the same perpendicular distance from the piano wire as every other molecule on that line. The lines which are bunched close together therefore represent compression waves, while those spread out represent waves of rarefaction.

Notice the wave form shown directly below the picture of the sound waves. This wave form is constructed from the picture by plotting the air pressure or air density at various distances from the piano wire, at one parmeans that sound is a variation of pressure in the transmitting medium.

► To put this important fact another way, we can say that a diaphragm struck by a sound will be alternately pushed in by the compression waves and pulled out by the waves of rarefaction. The pressure on the face of the diaphragm will vary above and below normal.

► Actually, a sound wave is very seldom a pure sine wave. The vibrating piano wire we have already discussed would, in addition to the fundamental sine wave pictured, also give off several harmonics of the fundamental, and might have an extremely complex wave shape. You will learn more about this a little later in this lesson.

SPEED OF SOUND WAVES

The speed of sound waves in any solid or liquid material depends upon two things, the *elasticity* of the material and the *density* of the material. You know that a material can be made to stretch or compress by applying force; the amount of compressing or stretching (the decrease or increase in size) is called the strain. The more force it takes to strain a material a given amount without permanently changing its shape, the greater is the elasticity or springiness of this material.

Notice—this scientific definition of elasticity is just about the opposite of the way most people use the word. The average man would say, for example, that rubber is more elastic than steel. But according to the scientific definition of elasticity, steel is much more elastic than rubber, for it takes much more force to compress a piece of steel a given amount than it does to compress a piece of rubber the same amount. Keep the scientific definition clearly in mind, for that is what we mean when we refer to the elasticity of a material. As a general rule, the harder a material, the more elastic it is. The density of a material is its weight per unit volume; if bricks of identical size arc made up from various materials, the material in the heaviest brick will have the greatest density.

► The more elastic a sound-conducting material is, the faster sound waves can travel through it; the greater the density of a sound-conducting material, the slower will be the speed of sound waves through it. Both elasticity and density must be considered in determining whether one material will transmit sound faster than another. If two material objects have approximately equal densities, you can determine readily which will transmit sound



FIG. 2. By plotting the air pressure we would find that the sound waves travel away from the wire as alternate changes in pressure. The actual "plot" will show the wave form of the source. Thus, we get a sine-wave plot if the source produces sine-wave sounds. (A piano wire produces a more complex wave than that shown here.)

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waves the faster by comparing their elasticities. For example, sound waves travel faster through aluminum than through soft rubber, because aluminum has a higher elasticity.

The speed of sound waves differs for various materials and even varies considerably with conditions in the same material. With air, for instance, the speed of sound waves increases with barometric pressure and with temperature. Careful laboratory experiments have shown that, at a normal atmospheric pressure and at a temperature of 0° centigrade, the speed of sound waves through air is about 1,089 feet per second. (These measurements are made in still air, since wind naturally affects the speed of sound in air.) Since the speed of electromagnetic waves (radio and light waves) is 186,000 miles per second, it is easy to understand why you can see lightning a long time before you hear the thunder which it produces.

Sound waves travel through water at a rate above 4700 feet per second. Steel is about seven times as dense as water (which would tend to slow up sound waves), but its elasticity is so much greater than water that sound waves will travel through steel at about 16,300 feet per second, almost four times their speed through water.

REFLECTED, TRANSMITTED AND ABSORBED SOUNDS

In radio and in the allied fields of public address and sound movies, we have a great deal to do with sounds produced in air. Ordinarily, these sounds are produced in rooms or in confined places where walls, ceilings or floors prevent the sound from traveling out in all directions. When sound waves strike a material, they are reflected from the surface of the material, absorbed by the material, or transmitted through the material; these three factors are shown in Fig. 3.

Echoes. When sound waves are reflected by a flat surface which is quite far away from the source of sound, wemay hear echoes, which are distinct but weaker reproductions of the original sound. Echoes are heard in many large rooms or in small rooms having curved walls which reflect the sound waves back to their source in a concentrated form. Echoes can be removed from a room by changing its shape or making it smaller; this clearly is a job for the architect.

Reverberation. When sound waves are reflected repeatedly back and forth



FIG. 3. Three things can happen to a sound wave which hits the surface of a material, as shown here. The kind of material will determine which one (or more) of these effects will occur.

between the walls of a room in a manner similar to that shown in Fig. 4, we have reverberation. The waves which reach the listener after reflection (over paths 2, 3, and 4) mix with the waves heard directly (over path 1) to give a "blurred" sound. The blur is produced because the reflected waves take longer than the direct waves to reach the listener, since they have farther to go. Thus, a sound heard over path 1 is followed a fraction of a second later by the same sound heard over path 2, then by the same sound heard over path 3, and finally by the same sound heard over path 4. This means that sounds coming over path 1 will be interfered

with by earlier sounds coming over the other paths. This reduces the understandability of words. If a great many such reflection paths exist, it may take a considerable time for a sound to die out, so that all intelligibility is lost.

The time required for a sound in a room to decrease to one-millionth of its original intensity is called the reverberation period of the room. Auditoriums which have reverberation periods of between one and two seconds are considered to have good acoustic qualities. The reverberation period of a room can be reduced by placing sound-absorbing material on the walls and ceiling.

Sound-Absorbing Materials. The better the sound-absorbing material used, the less sound will be reflected or transmitted. A soundproof room must have good sound-absorbing surfaces which keep out external sounds and also absorb sounds produced inside. An open window is sometimes considered to be an ideal sound "absorber" because it allows the sound to pass out of the room: on the other hand, it also allows external sound to enter the room, so you can't soundproof a room by leaving all the windows open! A sound-absorbing material dissipates the sound energy by converting it into heat. There are two ways of doing this: 1, a porous material has tiny "pockets" such that when sound waves enter, their energy is dissipated by being reflected back and forth in the pocket; and 2, soft materials will "give" under sound pressures but do not have the elasticity to bounce back and "throw" the sound wave back as a reflection. Thus, energy is absorbed by the "cushioning" effect of the material. Solid dense materials are good sound transmitters, and therefore poor absorbers; soft, pliable, and porous materials like



FIG. 4. Reverberations are produced whenever sounds can come to a listener over more than one path. Thus, sound waves reflected from the walls may travel over paths 2, 3, and 4 to the listener. As these paths are longer than the direct path (1), these reflected sounds arrive later than the direct sound waves. If they are of sufficient strength, they may blur the sounds. However, a certain amount of reverberation is desirable when listening to music.

velvet, Celotex, rock wool, cotton, carpet, and porous plaster are good soundabsorbing materials. The thicker the material, the more absorption of sound there will be.

▶ The fact that an audience absorbs sound waves is recognized by public address technicians; they reduce reverberation and echo effects by using directional loudspeakers which are directed at the audience rather than at the walls of the auditorium. The same thing is done in movie theatres-possibly you have noticed that the sound changes in quality as the size of the audience changes.

► So far, we have shown how sounds are produced, and also indicated some of the problems encountered when sound waves are produced within rooms. Reflection and sound absorption have much to do with the characteristics of the waves which we intend to pick up. Before we go into a study of pickups, however, let us learn more about the wave shapes which we will encounter.

Technical Facts About Sound and Hearing

A vibrating body, as you already know, produces sound waves. Since these sound waves consist of compressions and rarefactions of the particles in the transmitting medium, sound waves are studied best by measuring the pressure which they exert in the medium. Thus, in Fig. 2, the pressure which the sound waves (produced by a vibrating piano wire) exert at any point varies in exactly the same manner as the motion of the wire. If the diaphragm mechanism shown in Fig. 5A is placed in the path of a sound wave, the variations in sound pressure will bend the diaphragm in and out. This motion will be traced on a moving strip of paper by the link and pencil mechanism. If the sound wave is of a simple sinusoidal form, the tracing on paper will resemble the sine wave shown in Fig. δB ; if it is of a complex form, the tracing may be somewhat like that shown in Fig. 5C.

You will learn shortly that a complex sound wave, like that shown in Fig. 5C, consists of a basic or fundamental frequency and many overtone or harmonic frequencies, each of which can be considered to be a simple sine wave. Any simple sine wave has two important characteristics—amplitude and frequency—whether it is a smewave sound traveling in a sound-conducting medium or a sine-wave current flowing in an electrical circuit. The amplitude of a sound wave determines the loudness of the sound, while its frequency determines its pitch.

HARMONICS

Speech, music, and noise are never pure sine-wave sounds; they always consist of a fundamental frequency and many higher frequencies which we call harmonics or overtones. The harmonics or overtones give certain characteristic qualities (called "timbre") to a sound. That is, these combinations of fundamental and harmonic frequencies give to speech and music distinguishing characteristics which, through experience, we are able to interpret. Thus we are able to distinguish between a harmonica, a trumpet, a violin, a flute, and a tenor singer, due to the different harmonic values even though all may be producing the same fundamental frequency.

From the foregoing, you can see that any sound has three characteristics: loudness, pitch, and quality or timbre.

How Common Sounds "Look." The wave forms of a number of common sounds are illustrated in Fig. 6. These waves can be seen on the screen of a device known as a cathode rav oscillograph, which you will study later. This device would trace the wave shown at A, if you were to say "ah" before a microphone which was suitably connected to the oscillograph. A greatly different wave form, that shown at B, results when the same sound "ah" is sung. The key of C nearest the center of a piano keyboard (known to musicians as "middle C") produces the wave form shown at C. while street noise gives the very jumbled wave form shown at D. You will notice that with the exception of noise, wave forms of sound appear to repeat themselves at regular intervals; the time of one such interval determines the basic pitch or fundamental frequency of the sound.

► Musical tones, whether produced by stringed instruments, wind instruments, or the singing human voice, all consist of a fundamental frequency and a number of harmonics. For example, one of the C notes on a piano has a fundamental frequency of about 517 cycles per second. This note also contains a second harmonic of 1034 cycles, a third harmonic of 1551 cycles, a fourth harmonic of 2068 cycles, and a fifth harmonic of 2585 cycles. Each harmonic is a pure sine-wave tone.

The amplitudes of these harmonics differ greatly. The second harmonic in



FIG. 5. The simple mechanism at A can be used to trace sound waves, thus proving that sound waves exert a varying pressure on the diaphragm. Typical tracings are shown at B and C.

the above example is 20% of the amplitude of the fundamental, the third harmonic 25% of the fundamental, the fourth harmonic 10%, and the fifth harmonic about 8% of the fundamental. Another instrument producing the same note also may produce the same harmonics, but the amplitude of these harmonics would not be the same as those produced by the piano. In certain instruments, some of the harmonics are missing and, in some cases, a higher harmonic is stronger than a lower harmonic, or it may be even stronger than the fundamental frequency. It is the harmonics which make musical tones pleasing, while their number and amplitude determine the "timbre" or the characteristic which lets us distinguish between instruments.

► Speech differs from music essentially in that it is less melodic, being produced more nearly as a monotone. Speech sounds, like music, contain a fundamental pitch which distinguishes between the voices of children, women, and men, plus many different higher frequencies with certain frequencies predominating to give the characteristic distinctions between the voices of different persons.

▶ Noise is somewhat harder to define. Essentially, any unpleasant sound can be called a noise, whether it be just an excessively loud sound or whether it be musical notes combined in an unpleasing manner. Ordinarily, however, a noise is considered to be any irregular. distracting or unpleasing sound which would tend to mar musical reproduction or reduce the intelligibility of speech. Graphs of most noise sounds show irregular, sharp peaks, and are of an unsymmetrical nature. Noise contains overtone frequencies which are not necessarily harmonic frequencies-that is, they are not related mathematically in the same manner as are harmonics. Furthermore, instead of having a single well-defined fundamental frequency, the noise pulse usually will consist of a number of frequencies over a rather wide band.

FUNDAMENTAL FREQUENCIES

The piano, organ, and harp produce the greatest ranges of *fundamental* fre-

quencies—from about 16 to about 4096 cycles. A baritone singer produces fundamental frequencies between about 80 and 400 cycles; a piccolo can "tweet" between about 500 and 5000 cycles; a ukulele has the limited frequency range of between about 300 and 1000 cycles. Thus, each instrument has its own range of fundamental frequencies as well as harmonics or overtones of these fundamentals. Radio apparatus must handle all these fundamentals and the audible overtones, if it is to give high fidelity reproduction.

THE HUMAN EAR

The human ear is by no means an ideal sound-interpreting device. One can control his ears, causing them to hear desired sounds and to disregard others. This is helpful in ignoring noise, but the same mechanism serves to fool one into hearing sounds that are not present! Thus, a person familiar with the timbre produced by, let us say, a piano, will automatically "hear" this timbre although a radio may be reproducing only a part of the harmonics necessary for a true reproduction. This extra effort is tiring, however, and soon one does not wish to listen further.

► In addition, human ears actually can become accustomed to some types of distortion and like it. A boomy radio receiver, with an excessive lowfrequency response—sounding as if it were in a barrel—gives distorted reproduction, but many persons like this. In fact, some prefer it to high-fidelity reception. To musicians, however, or to others who appreciate high fidelity, any such distortion may be very annoying.

► Radio technicians should understand this queer behavior of the human ear; it is likewise of vital importance to the designer of radio apparatus. There is no need of spending time and money in reducing distortion when the improvement in fidelity cannot be noticed by the average human ear; nor is there any need to make a receiverrespond to frequencies which cannot be heard by the normal human ear.

By studying the responses of thousands of persons to sounds of various





frequencies, the Bell Telephone Company has found that the maximum range of frequencies which can be heard by even the best ears extends from 20 cycles to 20,000 cycles; others claim that this range is from 32 to 16,-000 cycles. These tests also showed that the human ear is far more sensitive to sounds in the 1000- to 4000cycle range than to sounds outside this range. Very high sound pressures are required before the ear can detect very low and very high frequencies in the audible range.

Sounds Can Cause Pain. When sound pressures are increased too high, a person stops hearing and actually begins to feel the sounds. At the lower frequencies, the vibration of the air can be felt by all parts of the body, but at the high frequencies, the action is more a sensation of pain in the ear. The response characteristic of the average human ear, obtained by noting when each frequency can just be heard and just be felt, is given in Fig. 7. A pure sine-wave sound which could be varied in frequency from zero to 20,-000 cycles was used in this test; the loudness of the sound was measured in terms of the pressure exerted on a flat surface. You will note that at 2000 cycles it takes about .0005* bars of r.m.s. sound pressure to make the sound audible, and about 1000 bars (2 r.m.s. pounds per square foot) before the sound can be felt. At this frequency, then, the pressure at the threshold (the beginning) of *feeling* is about 2,000,000 times the pressure at the threshold of audibility. This difference in threshold values decreases for higher and lower frequencies, as you can see in Fig. 7.

The Decibel. You might think that a 2000-cycle sound at the threshold of feeling should seem 2,000,000 times

louder to the human ear than a 2000cycle sound near the threshold of audibility, but this is not exactly the way in which the human ear responds. The amount by which a pure sine-wave sound must be increased before the change in sound level can be distinguished by the average human ear is called a "docibol." Doubling the pressure of a sound causes a 6-decibel increase in the ear's sensations. Increasing the pressure gradually to ten times its original value results in twenty sensations of increase in sound, or a 20decibel (abbreviated 20 db) increase. Increasing the pressure one hundred times results in a 40-db increase; increasing the pressure one thousand times gives a 60-db increase; and an increase of 2,000,000 times corresponds to about 126 db. Thus, you can see that increasing the sound pressure of a 2000-cycle note 2,000,000 times gives the equivalent of 126 separate and distinguishable increases in sound, as far as the human ear is concerned.

► The foregoing statements bring out a number of important characteristics of the human ear. Notice that doubling the pressure of a sound causes a 6decibel increase in the ear's sensation. This does not mean that doubling the pressure results in a sound six times as loud—it means that, as the pressure is gradually raised, there would be six recognizable increases in the volume. In other words, as the sound is increased gradually, there would be a certain point at which it would be possible for one to recognize that the new pressure produces more sound than the original pressure. Then, taking the new pressure as a basis, a further increase eventually would produce another recognizable volume step. There are six of these steps involved when the pressure is doubled.

This fact at once shows that the hu-

^{*}The bar is a term expressing pressure per unit area; one bar equals .002 pound of pressure per square foot of surface. Since sound pressure on any material is varying continually, we must deal with the r.m.s. value of the pressure, just as we deal with r.m.s. values of current and voltage. Sound or acoustical power is proportional to the square of sound pressure, just as electrical power is proportional to the square of voltage (electrical pressure). The acoustic bar is used here; one bar is equivalent to one dyne per square centimeter when the pressure is measured in metric units.

man ear is far more sensitive to pressure changes when the original volume is low. In other words, doubling the sound pressure from one bar to two bars results in a 6-decibel increase. and so would a doubling of pressure from one hundred bars to two hundred bars. Therefore, if the original sound level is low, a small change in pressure will produce a recognizable increase in loudness. On the other hand, the ear becomes less sensitive as the sound pressure increases, so that at high levels large changes in pressure are necessary before the ear can detect any change in volume.

into account. The amount of power fed to the loudspeaker determines the sound pressure the speaker will be able to develop and, in turn, the apparent loudness of the sound. Therefore, it is convenient to speak of the audio power in terms of decibels rather than electrical watts. This allows the designer to see at once just how much an increase in electrical power output will increase the sound output.

Again, a reference value is necessary. Any value can be chosen arbitrarily; radio men normally use the power level of 6 milliwatts, and call this the zero level or zero db. Notice—



FIG. 7. The response of the average human ear to pure sinewave sounds of various frequencies is given here. Sound waves having pressures within the shaded area can be heard. The r.m.s. pressure of the sounds is expressed here in "bars"; one bar equals .002 lb. per square inch.

Notice further that the decibel is a unit of sound change. It always represents the comparison between one sound level and another. Furthermore, it represents the number of steps in the increase. We cannot say that one sound is just twice as loud as another—but we can say that one sound is a number of steps louder than another.

POWER LEVELS

Because the ear responds to sound in the peculiar way just described, the radio designer must take this factor this is a unit used in radio when comparing electrical powers, and does not have any relationship to the acoustic reference level. In fact, we cannot say just how much acoustic power will be produced by this amount of electric power, as this depends on loudspeaker efficiencies, the acoustic treatment of the loudspeaker enclosure and a number of other factors. However, we can reasonably expect that a rise of 5 db in electrical power will produce a rise of 5 db in sound power.

From the foregoing, you can see that

the 6-milliwatt level is purely an arbitrary reference level, to which other electrical powers can be compared. Fig. 8 gives db values for various amounts of power in watts, when this reference level is used. You can see that a power level of 19 watts corresponds to about 35 db, 6 watts to 30 db, 1.9 watts to 25 db, etc.

If the power level is less than the reference value, the output is considered to be a minus decibel value. Thus, an output of .002 watts is about 5 db below the zero level (commonly called 5 db down, or minus 5 db, and written -5 db).

From the table, you can see that doubling the *power* results in a 3-db increase; multiplying the power by 10 causes a 10-db increase; multiplying by 100 causes a 20-db increase and so forth. Thus, the decibel increases for *power* changes are different from those caused by *pressure* changes.

► Regardless of this, the response of the human ear to sounds follows the decibel scale. As we have said, the average car can distinguish a 1-db change in sound if the sound is a pure sine wave. If, however, the sound is complex (speech or music for example) the smallest change in level which the average ear can notice is about 3 db (corresponding to a power ratio of 2 to 1).

From this, you can see that giving a power increase or decrease in terms of decibels lets us tell at once whether the increase or decrease will be noticeable to the ear. Thus, if we consider a complex sound, we have to increase the power three decibels, or double it, before the ear is able to tell that the sound has been increased at all. Hence, if an amplifier is feeding 6 watts to the loudspeaker, we must go to 12 watts before the increase would be noticeable. In other words, under the same conditions and for the same loudspeaker characteristics and efficiencies, it is not possible to tell the difference

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WATTS	DB	WATTS	DB	W	ATTS	DB
.076	11	.95	22		11.9	33
.095	12	1.2	23		15.2	34
.12	13	1.5	24		19.0	35
.15	14	1.9	25		23.7	36
.19	15	2.4	26		30.4	37
.24	16	3.0	27		38.0	38
.30	17	3.8	28		47.4	39
.38	18	4.7	29		60.0	40
.47	19	6.0	30		75.9	41
60	20	7 6	31		94.9	42

FIG. 8. This table shows the relation between electrical power (in watts) and decibels, when the zero db level is set at .006 watt.

between an amplifier producing 6 watts and one producing 7 watts.

► As we will see shortly, the fact that the human ear has this peculiar characteristic is quite helpful since it means that a form of distortion known as frequency distortion is not as noticeable as it would be otherwise.
Sound Pickups and Reproducers

We have shown that one of the important characteristics of a sound wave is that it exists as a variation in pressure. Therefore, it is obvious that a simple and direct means of converting a sound wave into an electrical current is to use a pressure-actuated pickup unit, so designed that it converts the variations in pressure into electrical current variations.

By the same reasoning, the sound reproducer must change electrical current variations into sound pressure variations.

Let us turn now to typical pickups and reproducers, and learn something of the fundamentals of their operation. At this time, we won't go into all of their characteristics — we'll just see how they operate.

SOUND PICKUPS

There are a number of different kinds of microphones in use today, but the five most common types are shown in Fig. 9. All of them use some form of diaphragm upon which the sound pressure can act. They differ principally in the method of converting the diaphragm movement into electrical current variations.

Carbon Microphone. Fig. 9A gives a simple sketch of the single-button carbon microphone. Part D, a very thin aluminum disc or diaphragm (about .001 inch thick) is stretched over a fixed metal ring R. The pressure of sound waves on disc D moves the disc back and forth, alternately squeezing and loosening carbon particles C in the telescoping metal sack or button labeled K. Carbon itself is a resistance material, but this device depends upon contact resistance for its operation. That is, as the particles are squeezed together, more of their surfaces will be in contact, so the resistance between them decreases. Then, when the particles are allowed to separate, they make poorer contact so the resistance increases. The result is that the electrical resistance between disc D and container K varies continually with the motion of the diaphragm. When this single-button microphone is placed in an electrical circuit containing a d.c. voltage E, the current passing through the circuit will be varied by this changing resistance which, in turn, varies in accordance with the wave form of the sound. Audio transformer T is placed in this circuit to transfer the variations in current to another circuit.

Condenser Microphone. Fig. 9B shows a simplified cross-section view of a condenser microphone. The thin aluminum disc or diaphragm marked D, mounted on ring R, is placed about .001 inch away from the fixed heavy plate P, thus forming a simple twoplate air condenser. These two plates are connected into a circuit containing a high voltage d.c. supply E and resistor R. Varying sound pressures change the distance between P and D, thus changing the capacity of the condenser. Changing the capacity this way results in a change in the charge stored in the condenser, so that the current through the circuit varies when the microphone picks up sound, and a varying voltage which has the same wave form as the original sound is produced across resistor R.

Dynamic Microphone. The dynamic microphone shown in Fig. 9Chas a thin diaphragm D on which is mounted a light-weight coil of wire. This coil moves between the poles of a permament magnet when the action of sound moves the diaphragm. As a result, there is induced in the coil a varying voltage whose wave form is a reproduction of the wave form of the sound.

Velocity Microphone. The velocity microphone illustrated in Fig. 9D operates on much the same principle as the dynamic microphone, except that here sound waves move a thin crimped metal ribbon M back and forth through a magnetic field produced by a permanent magnet. A voltage is induced in this metal ribbon; this induced voltphragm D, so that movements of the diaphragm caused by sound waves will bend the crystals. Electrons which flow to the flat surfaces of the crystals under this bending are collected by metal collector plates P.

► The electrical output of microphones is very small so it is necessary to amplify this output many times. Therefore, at the transmitter, the microphone will feed into an audio amplifier designed to raise the signal power to the point needed for modulation. In public address systems, a similar amplifier is used to increase the



FIG. 9. Simplified diagrams illustrating the operating principles of five different types of microphones. Each converts changes in sound pressure into equivalent changes in electrical signals.

age is stepped up by the transformer.

Crystal Microphone. The bending or straining of a Rochelle salt crystal produces charges of opposite sign on the opposite faces of the crystal. The potential difference which exists between these charges is proportional to the strain. This principle is utilized in the crystal microphone illustrated in Fig. $\mathcal{9E}$, where two square crystals are mounted back to back to increase the electrical action. This square crystal unit C is clamped at three of its corners; its free corner is linked to diasound power to the level necessary to operate loudspeakers.

SOUND REPRODUCERS

Headphones and loudspeakers are the devices used for sound reproduction. Although there are a number of types of each of these, they operate on the principle of converting electrical current variations into variations in air pressure, corresponding to sound waves.

unit C is clamped at three of its corners; its free corner is linked to dia- modulated at the receiver, there exists

a varying current which we call an a.f. current. This current is the electrical counterpart of the sound wave as it was transmitted and as it has been modified by its passage through the

enough power at the output of the average demodulator to operate a pair of headphones satisfactorily, but an audio amplifier is necessary before a loudspeaker can be operated. Thus, there is a miniature counterpart of the transmitter audio amplifier used in the receiver for the purpose of increasing the audio power to the level required by the loudspeaker.

Essentially, both headphones and loudspeakers operate somewhat on the principle of the dynamic microphone in reverse. They utilize electric signals to set into motion a diaphragm or a cone, alternately compressing and rarefying the air in front of the reproducer to reconstruct the original sound wave.

Headphones. In the common headphone unit, illustrated in Fig. 10A, a thin flexible steel diaphragm is placed over the two poles of a horse-shoe magnet, and a coil having many turns of insulated wire is placed around each leg of this magnet. The audio current flowing through the two coils alternately increases and decreases the attraction which the permanent magnet has on the diaphragm, causing the diaphragm to move in and out, thus producing sound waves.

Magnetic Loudspeaker. If a high sound output is wanted, the reproducer must be able to displace a large amount of air, and so must have a large diaphragm or cone which can be moved appreciable distances. The balanced armature electroniagnetic reproducer shown in Fig. 10B has been used widely for this purpose. In this loudspeaker, a soft steel armature is pivoted be-

tween two sets of N and S poles which are parts of a powerful permanent magnet. Also surrounding the armature is a solenoid or coil which carries the audio current. This coil makes the radio up to this point. Usually, there is -ends of the armature alternately of op-



FIG. 10. Simplified diagrams illustrating the operating principles of three different types of sound reproducers. Each converts electrical energy into mechanical energy (motion), which in turn is used to produce sound waves.

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posite polarity. The ends of the armature, therefore, alternately move towards and away from the permanent magnet poles. This movement is mechanically relayed to the large paper cone so that the paper cone is pushed in and out, setting the surrounding air into vibration and producing sound which is a reproduction of the original wave form. The outer edge of the cone is designed in such a way that it can be attached to its ring-shaped supporting frame without preventing free movement of the cone in and out under the action of the armature.

Dynamic Loudspeaker. The dynamic loudspeaker shown in Fig. 10C is the most widely used sound reproducer, for it is capable of delivering high sound outputs when the cone is used with a baffle.* Here, a coil of wire is wound on a thin bakelite or paper tube which is attached directly to a paper cone (as shown) or to a metal diaphragm (not shown). The audio signal passing through this coil causes a varying magnetic field. This field in-

N

teracts with a fixed field which is produced either by a permanent magnet or by an electro-magnet. The result is that the coil is forced to move in and out of the fixed field, driving the cone and thus moving the air ahead of it.

The amount of movement depends on the field strengths. Hence, the "voice" coil is fed with considerable audio power, and the fixed field is made strong by making the separation between the north and south poles of the magnet as spall as possible.

A dynamic loudspeaker can handle from 3 to 50 watts of audio signal power (depending upon its size), while a magnetic loudspeaker can handle from 1 to 3 watts. A headphone unit can handle only about 50 milliwatts. ► You are going to study loudspeakers and headphones in much greater detail in later lessons. However, you can see now that the purpose of the audio amplifier is to increase the power up to the level necessary to operate the loudspeaker. It would appear that the amplifier at both the receiver and the transmitter should be rather simple in its design and operation. Still, this is not altogether true, as this amplifier is largely responsible for the fidelity with which the electrical sound signal is handled.

^{*}A baffle is a large flat surface, a cabinet, or a horn-shaped enclosure which serves to prevent waves from the back of the speaker cone from interfering with those from the front, and also acts to "load" the cone by coupling it to a larger volume of air.

Requirements of an Audio Amplifier

All amplifiers may be divided into two broad groups, high-frequency amplifiers and low-frequency amplifiers. High-frequency amplifiers handle the r.f. carrier signals (which may or may not be modulated), while low-frequency amplifiers handle only the intelligence signals. In this lesson we will learn something about the requirements placed on low-frequency amplifiers. Let's cover first the low-frequency amplifier used in sound radio receivers.

The low-frequency amplifier in a sound radio receiver is called an audio amplifier because the frequencies it handles correspond to audio or sound frequencies. As we have previously mentioned, the audio amplifier is the section in a radio receiver following the second detector or demodulator. A corresponding amplifier is used in the transmitter between the sound pickup device and the modulator, and similar amplifiers are used in public address systems, electronic musical instruments, electrical phonograph amplifiers, and many photoelectric control devices.

The basic requirement of any lowfrequency amplifier is, of course, to increase the signal level. We apply a signal to the grid and obtain a change in the plate current. In turn, this plate current flow through the load produces a varying load voltage. This load voltage variation is larger than the grid voltage variation, provided that the value of the load has been chosen properly.

► We now have our amplified signal, but is it *exactly* like the original grid signal? If we have a sine-wave signal, we want to have an enlarged sine-wave signal across the load—if we start with a complex signal, we want the output signal to be exactly like it. However, if the output wave is different in any manner from the input signal (except for being amplified), then the signal is being distorted. The sounds we obtain from the loudspeaker will not be exactly like those originating in the broadcasting studio when the amplifier (or the speaker, for that matter) introduces harmonics which were not in the original signal, or when the relative amplitudes of any of the harmonics in the signal are altered.

From this, the manner in which the signal level is increased is of importance, so we must see how it is possible for an amplifier to distort the audio signal. Before studying the three forms of distortion, let's see how complex waves are formed, so that we can see just what may happen to the signal within an amplifier.

1

WAVE FORMS

Any wave form differing from a pure sine wave contains one or more harmonics. For example, in Fig. 11, the wave at C can be obtained by combining the fundamental frequency Awith its second harmonic B. You can check this fact by combining the amplitudes along the dotted lines. Thus, adding the distance 1-2 of the fundamental (the distance from the curve to the line R-R) to 3-4 of the harmonic gives 5-6 on the resultant.

The distance 9-10 on the second harmonic must be subtracted from the distance 7-8 on the fundamental to get the resultant 11-12, because the second harmonic is going through the opposite phase to that of the fundamental wave at this particular moment. Thus, you can see that, when both waves are on the same side of the reference lines R-R, their amplitudes add. If they are on opposite sides, the smaller is subtracted from the larger to get the resultant. Incidentally, the resultant wave will be changed in shape if the amplitudes of the fundamental and the harmonic are changed.

▶ In Fig. 12, we show the result of adding a third harmonic to its fundamental. Again the same procedure is used in combining the fundamental wave A to the third harmonic B to get the resultant wave C. Thus, adding the distance 1-2 to 3-4 gives 5-6. Subtracting 9-10 from 7-8 gives 11-12.

▶ When other harmonics are added, the same procedures are followed. As shown in Fig. 13, adding a number of harmonics gives another and entirely



FIG. 11. Here is the result of adding a second harmonic and a fundamental wave. Any wave which is not a sine wave consists of a fundamental and one or more harmonics.

different resultant wave shape. Again the same procedure is followed in that the amplitudes are added and subtracted to give the resultant. Thus, along the dotted line, the amplitude 1-2 is measured, the distance 3-4 is subtracted, the distance 5-6 is added, the distance 7-8 is zero, leaving the resultant, 9-10.

► From the foregoing, you see that there is an infinite variety of resultant wave forms possible, depending upon the harmonics which are added to the fundamental and upon the amplitudes of the harmonics. (In addition, as we shall learn later, the phase of the harmonics will change the resultant wave shape.)

However, regardless of the shape of the resultant wave, we can always say that any complex wave is a combination of simple sine wave forms. This is the reason that we can study the operation of amplifiers by using simple sine waves—the complex waves which the amplifier must handle are made up from these sine waves.

► As a summary, we know that:

- 1. Any complex sound or signal consists of a fundamental frequency plus harmonics.
- 2. If the relative amplitudes of any of these components are altered, we have distortion.
- 3. If any components are added, which were not in the original signal, we have distortion.

We can now go on to study the kinds of distortion and their relative effects. We will see that practical amplifiers are far from ideal.

FREQUENCY DISTORTION

If one section or stage of a radio strengthens certain frequencies more than other frequencies, or removes some frequencies entirely, the resulting sound will not be like the original. This distortion is known as frequency distortion, because certain frequencies are favored more than others.

As an example, suppose we have a 1000-cycle fundamental note with a fifth harmonic at 5000 cycles. If the amplifier responds better to 5000 cycles

than it does to 1000 cycles, the amplitude of the fifth harmonic will be increased more than its fundamental is, which means that the resultant wave shape will be changed. On the other hand, if the amplifier has a lower re-_ sponse at 5000 cycles than at 1000 cycles, then the fifth harmonic amplitude will be reduced, which again means that the resultant wave shape is changed.

You can see that frequency distortion is produced because an amplifier does not have the same gain for all the frequencies it is supposed to handle.



FIG. 12. Adding a third harmonic to a fundamental produces the resultant shown here. Of course, the actual shape of the resultant depends upon the amplitudes of the fundamental and harmonic waves. However, the general shape is correct, and should be compared with the resultant shown in Fig. 11. Notice that odd harmonics (the third, fifth, etc.) produce a resultant with both halves exactly alike, while even harmonics (second, fourth, etc.) produce a resultant which does not have symmetrical halves.

This type of distortion is caused by the circuit parts (coils, condensers and transformers) associated with the vacuum tube in the amplifier stage.

► Earlier, we mentioned that the ear does not notice a power change of less than about 3 db (power ratio of about 2 to 1) when complex sounds are be-

ing heard, which means that frequency distortion is less troublesome than would appear at first glance. It is quite permissible to have radio apparatus amplify some frequencies almost twice as much as others, or to convert some frequencies into sound twice as well as others, for the human ear is not able to notice this amount of distortion. This characteristic of the human ear is very much appreciated by the radio designer, since it is quite difficult for him to make radio apparatus amplify very low and very high frequencies equally as well as in-between frequencies.

Fig. 14 shows a typical response curve for an audio amplifier. Notice that the scale at the left is in decibels, because this permits us to compare the response to the hearing capabilities of the ear.

► When making a curve of this kind, the power output at some reference frequency (usually 1000 cycles) is taken as the standard or reference value. Then, the power output at other frequencies is found. If the amount of output is less than that at the reference frequency, then the db output is less, while greater outputs give increases in the decibel level.

As you will notice from the graph in Fig. 14, the curve is within 3 db of the reference level (24 db) from about 80 cycles to above 4000 cycles. At 5000 cycles there is a peak in the response which is nearly 6 decibels higher than the reference, so the output at frequencies about this value would be noticeably greater.

Above 8000 cycles, the response cuts off sharply, so frequencies above this point are poorly reproduced, if at all. ► When the very high frequencies are cut off by radio apparatus, sound loses some of its fidelity of reproduction. However, not all the high frequencies up to 20,000 cycles are necessary for understandability or appreciation of music. You can cut off sounds at about 8500 cycles and still get high quality speech and music. Why reproduce frequencies up around 15,000 to 20,000 cycles, when only very young people and trained musicians are able, as a rule, to hear these frequencies at all?



FIG. 13. When a number of harmonics are added, the resultant wave shape depends upon whether the even or the odd ones predominate, and upon the amplitudes of all of them.

With age, one's hearing ability decreases so that elderly people cannot even hear frequencies above 6000 cycles in most cases. Therefore, for all practical purposes, it is not necessary to reproduce these high frequencies.

► Most of the power needed to reproduce music is used to reproduce low-frequency notes, and, since the ear is not sensitive to the lows and highs (low and high frequencies), the reproduction of sound at a volume much below that of the original makes the music sound unreal. It is for this reason that the so-called low (or bass) boosters are included in radio receivers to amplify excessively these lower frequencies and get effects which sound more natural to the ear. Some of these boosters are automatic compensators, while others are tone controls, adjustable at the will of the customer. Thus, it is quite possible that the customer will even introduce a certain amount of frequency distortion, in order to make the sounds more natural to his ears!

Of course, if too many of the high or low frequencies are cut off, then the frequency distortion becomes objectionable.

AMPLITUDE DISTORTION

When a radio circuit or device does not produce current or voltage changes which are *exactly proportional*, at each instant of time, to the changes occurring in the voltage or current values of the incoming signal, we have what is known as *amplitude distortion*.



FIG. 14. Here is a typical response curve for an audio amplifier, showing the output over its frequency range. The output is given in db above the .006 watt zero level, so that variations in the response can be judged in terms of their effects on the human ear. Using the db table in Fig. 8, practice converting the db values into watts. Thus, 24 db is equal to 1.5 watts, etc. (Sometimes you will find that a similar curve is used, but the reference frequency output is labelled 0 db. Then values above are written as +db, while those lower are written as -db. That is, when 24 db is called 0, 27 db becomes +3, 30 becomes +6, 21 becomes -3, and 18 becomes -6 db. This allows us to consider the output in terms of the db change above or below the reference level.)

If the original wave is a sine wave, then amplitude distortion will result in the production of harmonics which were not present in the original signal. If the original signal is of a complex nature, additional harmonics are added, or the relative amplitudes of the harmonics are changed by amplitude distortion.

Fig. 15 shows a typical example of amplitude distortion. At A, we have a sine-wave signal. We want the amplifier to increase the amplitude of this signal, as shown at B, without changing its wave shape. However, if we get the amplified signal with the wave shape changed as shown at C, then we have amplitude distortion. As you can see, this wave is not an exact replica

that even harmonics, such as the second, fourth, sixth, etc., have been added to the fundamental, as shown in Fig. 17. In other words, the tube's action is approaching that of a half-wave rectifier. From your knowledge of rectifiers, you will remember that this wave form has a d.c. component as shown in Fig. 17. This d.c. current will be added to the normal d.c. current value. This means that, when a signal is applied to an amplifier tube operating on the lower bend of its characteristic, the resulting distortion will cause the d.c. plate current to rise. This is one of the means whereby a serviceman can determine when distortion occurs-he can apply a sinewave signal and measure the d.c. plate



FIG. 15. A typical sine wave is shown at A. When it is amplified, it looks like that shown at B. However, should amplitude distortion occur, it may look like the wave shown at C.

of the original sine-wave signal.

▶ What could produce this change in wave form? One of the most common sources of amplitude distortion is the operation of a class A amplifier over a curved tube characteristic, as shown in Fig. 16. Notice that the input signal $e_{\rm g}$ has more control over the plate current on its positive swing than it does on its negative swing. This results in an output signal (produced by plate current $i_{\rm p}$) which is not a sine wave or rather, it is a pure sine wave plus harmonics. Thus, although we started with a sine-wave signal on the grid, we are obtaining a sine wave plus harmonics in the plate circuit.

Analyzing the wave form of the output signal shown in Fig. 16, we find current before and after the signal is applied. If the stage is a class A amplifier, there should be no change in the d.c. plate current. If there is a change, distortion is taking place.

Should a complex wave be applied to the amplifier having the characteristic shown in Fig. 16, even harmonics will be added to the wave if they are not already present, or the amplitudes of any even harmonics which are present will be increased by this distortion.

► Going further, a similar distortion can occur if the grid is allowed to swing positive. As shown in Fig. 18, there is an upper bend in a tube characteristic; this will also produce distortion. The output wave is like that of Fig. 16 except that it is reversed, or



FIG. 16. Amplitude distortion is produced whenever the E_z - I_p curve of a tube is not linear (straight) in the operating region.

is upside down. It has the same components as are shown in Fig. 17, except that they are shifted in phase 180°. The negative (downward) swings are the larger, so the d.c. component will be of the opposite polarity to that shown in Fig. 17. In this case, there will be a reduction in plate current when the signal is applied.

▶ Therefore, if the d.c. plate current changes when a signal is applied to a class A stage, distortion is occurring. If the d.c. plate current rises, the tube is operating on the lower bend of its (This characteristic. operation is caused by excessive bias or low plate supply voltage.) If the d.c. plate current falls, the tube is operating on the upper bend in its characteristic. (Operation here is caused by too little bias or by an excessive plate supply voltage.) Notice the value of knowledge to the serviceman. Upon observing that distortion is occurring, he can take a meter reading and, from his knowledge of circuit actions, he can reason directly to the probable causes of the trouble!

► There is another kind of amplitude distortion in which odd harmonics are added instead of even harmonics. This distortion will be produced when the signal applied to the grid is excessive, so that operation occurs over *both* tube curvatures as shown in Fig. 19. Here, the shape of the output wave indicates that the third, fifth, seventh, and other odd harmonics are being added to the fundamental. This distortion is very noticeable to the ear, but it will not cause a change in the d.c. plate current if both halves of the output wave are exactly alike. The serviceman must use a cathode ray oscilloscope to actually "see" the wave form before he can find the stage producing this diffi-



FIG. 17. Here is an analysis of the wave produced by the E_{g} -1_p curvature which was shown in Fig. 16. Notice that even harmonics are added, plus a d.c. component. Although we have shown only the second and fourth harmonics, there are actually small amounts of other even harmonics added. A phase shift is also involved in that the harmonics are added 90° out of phase wit!. the fundamental. This phase shift cannot be heard, but the added harmonics are noticeable. (In television, the phase shift is of importance also.)

culty, unless there are other clues in addition to the distortion.

► Incidentally, it is possible to tell from the wave shape whether the distortion is an even or an odd harmonic addition. If the two halves of the resultant wave are *exactly* alike, only odd harmonics have been added. If two halves are not symmetrical, however, then even harmonics have been added. Where both even and odd are added, it is not so easy to determine this unless one or the other happens to predominate.

Amplitude distortion is readily detected by the ear. Strangely, the ear seems more critical of odd harmonic distortion than it is of even harmonic distortion. An odd harmonic distortion causing a change in harmonic levels of more than 5% begins to be noticeable,



FIG. 18. Amplitude distortion will occur also if the tube is allowed to operate at the upper bend of its characteristic.

while an even harmonic distortion of as much as 10% is relatively unnoticeable.

PHASE DISTORTION

When a radio circuit or part changes the phase relationship between different frequencies in the signal, we have phase distortion. The distance (phase) between peaks of the different signal components is changed, with the result that there is a change or distortion in the wave form of the output signal. (Later in this lesson we will show an example of phase distortion.)

The human ear is relatively unconscious of even large amounts of phase distortion, being far more critical of amplitude or frequency distortion. The reason for this is that the ear hears the individual components of the wave rather than the resultant wave. That is, the sound wave is "taken apart" by the ear. Hence, the ear is quite conscious of added or deleted frequencies, and of changes in amplitude. It cannot detect changes in *time* in the arrival until such change approaches the time length of a syllable or of a musical note.

However, phase distortion is far more important in television. As we shall learn later in this lesson, the time delay caused by even a small phase shift in a television signal will be noticed readily by the eye.

AUDIO AMPLIFIER REQUIREMENTS

From the foregoing, we can begin to see just what the requirements are for an audio frequency amplifier. In order, we may well list them as follows:

Voltage Amplification. A certain amount of voltage amplification is necessary in all radio receivers, as the voltage output of the average second detector is too low to operate the power output stage. Therefore, one or more voltage amplifying stages are used between the second detector and the power stage.

Power Amplification. Raising the voltage level is not enough—*power* is required to operate a loudspeaker. Therefore, the output stage of the audio amplifier in a radio receiver will

always be a power stage, designed to deliver sufficient power to operate the loudspeaker. Actually, this stage can be said to be converting signal voltages into high signal powers.

As you have learned in earlier lessons, one of the important differences between power stages and voltage amplifying stages lies in the choice of the



FIG. 19. When the input signal is allowed to become so large that it exceeds the normal operating region, both halves of the output wave will be distorted. This shows that odd harmonics have been added. Strangely, this distortion is far more noticeable than is even harmonic distortion.

load values. However, there are other important differences as we shall soon learn. In public address systems, where large amounts of signal power are required, medium power stages may be used to drive high power stages. Similarly, in transmitters, the power levels may be quite high. In these two instances, true power amplification does occur, for the final power stage is actually *driven* by a low power stage.

Distortion. From the foregoing, you can see that every effort must be made to cause tubes to operate on the straight-line portion of their characteristic curves. Inductance and capacity values must be carefully chosen and controlled to prevent the undesirable formation of low-pass and highpass filters which would tend to limit the frequency response. Even a matter like the choice of the load value is a compromise between the amount of amplification desired and the fidelity of response wanted.

▶ In your next lesson, you will learn much more about these important requirements and about how they affect the serviceman and his choice of replacement parts.

25

The Fundamentals of Television

Television involves the transmission of intelligence which reaches our brain through our eyes. First, let us consider what the eye sees when it looks at an object. Ordinarily, it looks at reflected light, made up of electromagnetic waves; occasionally, it looks directly at light sources such as electric lamps, a fire, or the sun. The eye sees color because the electromagnetic waves in the visual band have different frequencies, each frequency or group of frequencies giving, through the action of the brain, a color sensation. The human eve serves as a complicated lens (much like the lens in a camera), for it projects these electromagnetic waves on the retina, a surface at the back part of the eye. This retina is composed of millions of nerve ends, each of which is connected to the brain. These nerve ends interpret the strength of each electromagnetic wave which hits them (determined by the brightness of the object) and they also interpret the frequency of the wave (the color of the object), Each nerve end "sees" only a tiny portion of the entire scene: the brain reconstructs the over-all picture by assembling all the nerve impulses. Thus, the eye breaks up the scene into elements. each of which is transmitted over a separate nerve channel to the brain.

One scientist calls the human eye nature's own television system. The object viewed acts as the transmitter in the system, sending out electromagnetic waves which are picked up by the eye acting as a receiver, and then are relayed to the brain to give us the sensation of seeing.

A Suggested Television System. This action of our visual mechanism immediately suggests a method of con-

structing a television system. Why not arrange thousands or millions of tiny electric eyes on a screen to pick up the light waves, and connect these by thousands of wires or radio frequency transmitters to a receiver containing thousands of tiny glow lamps? Each of these would reproduce the amount of light picked up by its corresponding electric eye, so the combination of all the lamps would reproduce the object viewed by the transmitter. Yes, a television system like this has actually been tried for land wire television, but only on a small scale. The scheme was found to work after a fashion, but obviously was far from practical, as entirely too many wires were necessary.

Practical Television Systems. The television systems in use today do not attempt to pick up a complete scene and transmit it to a receiver all at once. Instead, television takes advantage of an eye characteristic known as <u>persistence of vision</u>—the ability of the eye to retain an impression of an object for a short time after the object has disappeared from view. This makes it possible to send a portion of a scene at a time, so long as the entire scene is transmitted before the eye "forgets" the first of it.

The scene is broken up into elements by scanning or by viewing a small portion at a time. Scanning is an operation very like what you do when you read this page. You don't look at the page and attempt to read every word at once. Instead, you read the first line from left to right, swinging quickly back to the left-hand side of the second line, read the second line, go back to the start of the third, and repeat the process until you have taken in every word by itself.

That is just about what a television "camera" does. (This camera is the pickup device of a television system, corresponding to the microphone in a radio system.) In effect, an "eye" in the camera travels over the top edge of the scene from left to right, swings quickly back to the left-hand side, moves down slightly, travels horizontally over the scene again, and repeats the process until the whole scene has been scanned. As you no doubt know. or have guessed, this "eve" is really a light-sensitive surface which converts the light received from the scene into an electric current. This current, which of course varies as different parts of the scene come into view of the scanning eye, is then transmitted by radio



FIG. 20. The drawing at A is reproduced as a series of lines at B and C. Greater detail is obtained by using more lines, as at C.

to the receiver. At the receiver, the process is reversed, and the original scene is traced out line by line.

This is a highly simplified version of how a television system works, but it will serve to show you the basic idea of operation. Right now, the important fact for you to grasp is that a scene is televised "bit by bit," not as a whole.

 \blacktriangleright Fig. 20 illustrates the general effect

produced when a scene is scanned. Suppose we wish to televise a picture like that in Fig. 20A. After it has been scanned by the camera, transmitted to the receiver, and reproduced on the receiver screen, it will have something of the appearance shown in Figs. 20B and 20C. That is, it will consist of a series of lines; these lines will vary in brightness along their length, and so make up the picture we see. The more lines we have in a given area, the greater the detail of the final picture. Fig. 20C, which has 120 lines, exhibits more detail than Fig. 20B, which has only 60.

Note that, as you move the illustrations in Fig. 20 farther and farther away from you, a point is reached for each illustration where the details seem to blend into a complete and nearly perfect reproduction of the original. This brings out an important fact about television: if a reproduced picture is made larger without increasing the number of lines, the picture will have to be viewed from a greater distance to get a satisfactory eye impression.

HOW SCENES ARE SCANNED AND REPRODUCED

Before considering the technical details of breaking up a scene into a number of lines, it will be valuable to get clearer ideas of how a scene is taken apart or *scanned*, and how a scene is reproduced. Only the basic schemes will be considered; naturally, different television experts have different ways of accomplishing the desired results.

Mechanical Scanning Methods. Even though mechanical methods of scanning are considered inadequate today, we will consider them first since they are easier to understand. Punch a hole in the center of a small business card with a pin and hold the card up to one of your eyes, so you can look

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through the hole. Turn to some object or scene. Notice that you can see only a small part of this scene through the tiny hole. Now move the card horizontally from left to right; you see all the portions of the scene along the line which you are scanning. Move the card back and forth horizontally while shifting it vertically downward a lit-



FIG. 21. This diagram shows an elementary mechanical scanning disc system. If the disc is rotated rapidly enough, the observer will be unconscious of its presence, as persistence of vision will allow him to apparently see the entire scene, although he is actually viewing only a tiny spot at a time.

tle at the end of each line, and your eye will scan the entire scene, piece by piece.

The Scanning Disc. In place of this crude scanning device, we can use the system shown in Fig. 21A, in which a large number of holes are arranged in a spiral fashion on a rotating disc called the *scanning disc*. This disc really replaces the business card used in our previous example: one complete revolution of the disc gives one complete scanning of the entire picture, for each hole on the disc scans one line. If the disc is revolved fast enough, the visual sensation is the same as if the entire picture were being seen at one time.

The exact arrangement of the holes on the scanning disc is shown more clearly in Fig. 21B. The observer views the scene through the mask, a rectangular opening in a piece of black cardboard. As the disc is rotated, each hole moves across the opening in this mask, the outermost hole in the spiral moving across the top of the opening and each succeeding hole moving across one line down. Finally, when the innermost hole has moved across the bottom of the opening, the outermost hole again scans the top line and the entire scanning process starts over again.

Mechanical Television Transmitters. If the observer in Fig. 21A is replaced with a light-sensitive cell, this cell will deliver a varying electric current which is at all times proportional to the amount of light reaching the cell, and therefore proportional to the shade of lightness or darkness of the element of the picture being scanned at a particular instant. This arrangement gives us a means of converting a picture or scene into a varying electrical current. This current or picture signal can be amplified and placed on a radio carrier for transmission through space. At the receiver, the carrier can be demodulated and the picture signal amplified sufficiently to operate a picture reproducer.

Mechanical Television Receivers. In the early television receivers, the amplified picture signal was fed to a neon glow lamp like that shown in Fig. 22A. This lamp consisted of a wire anode and a rectangular flat metal piece (the same size as the reproduced picture) which served as a cathode. These elements were enclosed in a gasfilled glass envelope. A red glow of light formed on the plate when sufficient voltage was applied between the electrodes; the intensity of this glow varied with the applied voltage. The amplified picture signal was made to change the applied voltage, thus changing the intensity of the glow.

A pin-hole scanning disc was rotated before the glow lamp in such a way that the holes scanned the glowing plate. The transmitter and the receiver were so synchronized that when the scanning disc at the transmitter



FIG. 22. An early type of mechanical television reproducer. The glow-lamp light depended on the brilliancy of the spot being scanned at the transmitter at that moment. The scanning disc is synchronized (in step) with the transmitter disc so that it arranges the light elements in their proper sequence.



Courtesy Don Les Broadcasting System

The engineer is holding an electronic pickup tube such as is used in television studios. Scanning is accomplished within this tube by electronic means.

started to scan the top line of the scene, the receiver scanning-disc likewise started to scan the top line. Line by line the scanning discs were kept in step or in synchronism, so that the intensity of the glow lamp at any instant corresponded to the intensity of the light reflected from that same element on the actual scene. The arrangement of the scanning disc and glow lamp are shown in Fig. 22B. The lens shown is a magnifying glass used to enlarge the image to three or four times the size of the glow lamp plate.

Electronic Television Transmitters. Although present day methods of scanning and picture reconstruction differ greatly from the method just described, the principle of breaking up a picture into a number of elements which are scanned line after line is still

World Radio History

used. Figure 23 illustrates a modern electronic television camera. The scene is focused on the photoelectric plate by a high-grade camera lens combination. This light-sensitive photoelectric plate consists of millions of tiny light-sensitive spots, each insulated from the others and each scarcely larger than the point of a pin. Under a microscope this plate looks as if it were covered with grains of sand.

When a scene is projected on the photoelectric plate by the lens, the ac-

amount of light reaching that section. Thus, some spots on the plate are more positively charged than others, and we actually have an electronic image of the scene. An electron gun now shoots a fine stream of electrons at the photoelectric plate. Electromagnetic deflecting coils (here designated as "deflecting yoke") shift this electron beam horizontally and vertically, one line at a time, to scan the entire photoelectric plate from top to bottom. When this electron stream strikes a positively



FIG. 23. A cut-away sketch showing the arrangement of parts inside one type of electronic television camera.

tion of light drives out electrons from each of the tiny light-sensitive units. These electrons pass through the space in the tube to a conducting surface on the inside of the glass envelope, which is at a high positive voltage and therefore attracts the electrons. The action of light thus leaves the photoelectric plate elements more or less positively charged (because they have lost electrons).

Naturally, the amount of electrons lost from any given section of this photoelectric plate depends upon the charged surface, that surface recovers its electrons and, in so doing, relays the charge to a flat metal supporting electrode which is back of, but insulated from, the photoelectric plate.

In this manner, an electronic impulse is relayed from each spot which is hit by the electron beam. The size of each impulse corresponds to the amount of light striking the spot, so the sum of all the impulses (sent one at a time) constitutes a picture signal. The supporting electrode collects the picture signal and, after a great deal of amplification, the picture signal is placed on a carrier wave and transmitted through space, just as in the mechanical television system. In addition, impulses are sent at the beginning of each "frame" or new picture, to keep the image-reconstructing devices in step with the scanning mechanism at the transmitter.

Electronic Television Receivers. Figure 24 shows a simplified diagram of a typical electronic picture reconstructor. This employs an electron gun and two sets of electromagnetic deflecting coils. Special oscillators generate the current pulses which flow through these coils; the oscillators are controlled by the synchronizing impulses sent out by the transmitter. A spot of light appears on the special screen at the end of the tube when it is hit by the electron beam produced by the electron gun; the brilliance of the spot increases with the speed of the electrons in the beam and with the number of electrons in the beam.

The picture signal voltage controls the speed and number of the electrons in the beam by means of a special grid electrode, and the deflecting coils control the scanning of the beam across, and up and down, the screen. The combined action is such that while the beam is sweeping across the screen, its intensity is changing continually in accordance with the picture signal, and the effect of "painting" light on the screen is secured. The picture size is controlled here by the size of the screen; a 12-inch diameter tube gives a $7.5'' \times 10''$ picture.



FIG. 24. A simplified diagram of an electronic picture reconstructor tube.

Video Amplifier Requirements

Television, like sound systems, requires both low- and high-frequency_ amplifiers. However, the requirements of both are far more exacting than those on the corresponding sound amplifiers. Basically, there must be voltage and power amplification, but the very wide frequency band which must be handled with a minimum of distortion requires special tubes and unique circuits, as we shall point out in later lessons.

VIDEO FREQUENCIES

The low-frequency television amplifier (called a video amplifier) appears almost misnamed, as it must handle frequencies which are well up in the r.f. range of sound systems. However, as in sound systems, the video signal is only that which was removed from the carrier by the demodulator; the television carrier is far higher in frequency.

First, let's see just what range of frequencies must be handled by the video amplifier. To begin with, we want a picture with the greatest possible amount of detail. People are familiar with the details of photographs and the movies and naturally expect television to be as good. As technical problems are solved, details become better, so that today the fidelity is acceptable. Even so, still greater fidelity is being sought.

► Another problem is *flicker*. To give the sensation of motion, the scene must be scanned over and over, with one scanning rapidly replacing another, just as in the movies. Each scanning is called a "frame" and, the more frames there are per second, the less chance there is for the eye to see them individually—they blend together better. If too few are transmitted, a flickering of the image becomes noticeable. Greater picture detail can be obtained by increasing the number of lines per frame; increasing the number of frames per second gives less flicker of the reproduced picture. Both of these factors contribute to what is called high-definition (or high-fidelity) reproduction. However, there are definite limits to the number of lines and frames which can be handled.

Picture Elements. The maximum frequency of the picture signal current is of great interest to every television engineer, for all television equipment must be designed to handle this frequency. In order to have a way to calculate the maximum frequency of the picture signal current, it is assumed that the picture being scanned consists of a checker-board pattern of black and white squares, with each square being equal in size to one of the sensitized spots on the photoelectric plate of the television camera. (Since each of these sensitized spots is the smallest part, or element, that the camera can "see" of a scene, these spots are usually called "picture elements.")

The signal current is said to go through one cycle each time the electron scanning beam passes over one light and one dark picture element, because the signal current goes through a maximum and a minimum value each time this happens. For our checkerboard scene, the picture signal current goes through a cycle each time the scanning beam passes over two consecutive picture elements. To find the maximum frequency of the picture signal current, then, all we have to do is compute the number of picture elements scanned per second and divide by two (since it takes two elements to make one cycle). Let's go through the simple computations involved and see just how high this maximum frequency may be.

If each picture element is considered to be as high as it is wide, it is easy to compute the number of elements in one complete picture. Assuming a square picture with N lines, then there will be N picture elements per line, or N times N picture elements in the complete square picture, which is known technically as one frame. For example, at present the television standards call for a 525-line picture. Hence, in a square picture there will be 525 times 525 or 275,625 picture elements. For ordinary calculations, 276,-000 elements will be sufficiently accurate.

Aspect Ratio. The pictures commonly involved in television are not square, however; they are wider than they are high. The width of a picture divided by its height is called the aspect ratio. In order to conform to motion picture standards, the aspect ratio has been standardized at 4/3 or 1.33. This means that the number of elements in each line has been increased by the aspect ratio, which we will designate as a. Now the number of picture elements per frame or picture will be N times N times a. For the example just considered, the total number of elements will therefore be 276,000 times 4/3 or 368,000.

Frame Frequency. The number of pictures sent per second is the *frame* (or picture) *frequency*; let us designate it as F. By multiplying the number of picture elements in a frame by the frame frequency, we get the total number of picture elements per second. The total number of picture elements per second is then $N \times N \times a \times F$. Since it takes two picture ele-

ments to make a cycle, we get the maximum number of cycles per second by dividing the preceding formula by 2. The standard frame frequency is 30; in our example, then, we get the maximum frequency involved by multiplying 368,000 by 30 and then dividing by 2; the result is 5,520,000 cycles per second.*

Factors Affecting Frequency. In actual television practice, the picture elements are being scanned only about 85% of the time. The remainder of the time is used for sending line synchronizing impulses. This fact increases our maximum picture frequency because we must crowd our picture elements into 85% (85/100) of a second. We must, therefore, multiply our computed value by 1.17, making the maximum picture frequency 1.17 times 5,520,000 or approximately 6,458,400 cycles.

Up to this point, our analysis of the maximum frequency has been based upon the assumption that there is always a sharp contrast between adjacent elements of the picture or scene, one dark and the next light. Of course, this is not true in actual practice, for no scene is made up of perfectly arranged checkerboard squares. Several adjacent picture elements in a line may reflect the same or nearly the same amount of light. Also, in most scenes, especially those having action, it is not necessary that slight variations between the shades of adjacent picture elements be transmitted. The average scene thus contains considerably less than the maximum possible number of cycles (changes from light to dark).

This is quite fortunate, for it reduces the maximum frequency required. Tests and experience have

^{*}The formula now is: Maximum theoretical picture frequency $f_p = \frac{1}{2} N \times N \times a \times F$.

shown that apparatus capable of sending about 60% of the maximum frequency is fairly satisfactory. Since the maximum number of cycles was assumed in our example, we multiply 6,-458,400 by .6 and get about 3,900,000 cycles (3.9 megacycles) as the final maximum frequency for a 525-line picture scanned 30 times per second, with the standard aspect ratio of 4/3. Any increase in this frequency up to the



FIG. 25. The double-spiral type of scanning disc at A illustrates the principle of interlaced scanning. The arrangement is such that the successive holes on one spiral skip every other line of the image, and the holes on the other spiral scan the lines missed by the first spiral. The result is shown at B. Whether produced by the double-spiral disc at A or by an electronic television system, the lines of the picture are scanned (and reproduced) in the order 1-3-5-7-9-11-13-15-17-2-4-6-8-10-12-14-16-18.

extreme limit of 4.5 megacycles permitted within a television band gives a definite improvement in picture fidelity.

The Minimum Frequency. The upper part of the average outdoor scene (usually the sky) is bright, while the lower part is considerably darker. In scanning such a scene, the picture elements are varying in light intensity at a high level for the upper half of the picture and at a low level for the remainder of the picture, giving one cycle of change from light to dark for each scanning of the picture. Transmitting these changes properly calls for a minimum frequency corresponding to the vertical scanning frequency (the frame frequency). Satisfactory reproduction of slow changes in background illumination requires, however, that frequencies down to at least 10 cycles be passed, so we should consider 10 cycles as the minimum frequency for a practical high-fidelity television system.

► For a 525-line picture having an aspect ratio of 4/3 and a frame frequency of 30, the picture frequency ranges from a minimum of 10 cycles to a maximum of about 3.9 megacycles. Compare this with the frequency range of high fidelity sound, which extends from about 35 to 8500 cycles per second!

Interlaced Scanning. Increasing the number of frames per second reduces flicker, but it also steps up the maximum signal frequency. For example, a 525-line picture with an aspect ratio of 4/3 and 30 frames per second gives a maximum frequency of about 3.9 megacycles. At 60 frames per second, the maximum frequency becomes about 7.8 megacycles, which is way beyond the present ability of television apparatus. But scientists were not to be balked; they resorted to interlaced scanning, an old principle of television which was used on many of the scanning disc systems. In these systems, the holes were arranged in two sets of spirals, as shown in Fig. 25A. There was one set of holes scanning or reconstructing every other line, as illustrated in Fig. 25B. Although the number of lines in the picture is not increased, the frame is scanned vertically twice for each complete scanning of all the elements. This does not change the maximum signal frequency, but reduces flicker considerably. Each of the "scannings" are called a "field" to distinguish it from the complete image which is called a

frame. There are two fields in each frame, in the system just described. ► This same principle of interlaced scanning has been applied to electronic television methods. It was discovered that at a field frequency of 60, by using an odd number of scanning lines per picture, the picture would automatically interlace. This is why 525 lines is the present standard rather nals, phase distortion is of little consequence because the human ear is concerned only with the frequency of a sound and with its strength, but in television, the phase distortion may tend to shift an impulse from one picture element to the other which definitely will cause distortion.

The effects of phase distortion are illustrated in Fig. 26. When the signal



FIG. 26. A comparison between the resultant waves will show that phase distortion will produce a different wave shape. In television, this means the spot brilliancy at the receiver will not correspond to that of the scene at the transmitter.

than 500 lines or 520 lines or any other even number.

PHASE DISTORTION

Radio apparatus designed for sound signals must handle the required range of frequencies in the sound, and must not introduce either frequency or amplitude distortion. Television amplifiers also have these requirements, as well as one other, that there be negligible phase distortion. With sound sigconsists of a fundamental frequency A and a third harmonic B, the resultant wave C will cause, when fed into a suitable reproducing device, light and dark areas to appear on the screen in exact reproduction of the light and dark areas of the original scene. If, however, radio apparatus were to shift the third harmonic by 90 degrees, as at D and E, the resultant wave at Fwill no longer represent the original scene. This can easily be proved by comparing corresponding points, such



Courtesy General Electric Co.

A combination television and all-wave receiver. The television reproducer tube is at the upper left and the television receiver controls are in the center. The broadcast-short wave receiver controls are at the upper right. The loudspeaker (used for both the all-wave set and for the television sound program) is behind the grille cloth at the bottom of the cabinet front.

as x, on the waves C and F. Since the amount of light produced on the screen of the image reconstructor tube depends on the amplitude of the wave, this difference in amplitude may easily make the image element far lighter or far darker than the original.

Looking Ahead. Now that you have some idea of the requirements placed on low-frequency amplifiers, you are ready to study practical amplifiers. In your next lesson, you will learn how low-frequency amplifiers are made to meet the requirements you have just studied. Then, in other lessons, you will go on to high-frequency amplifiers, demodulators, and other radio sections and stages. Soon you will have a complete understanding of *all* the sections in modern receivers.



World Radio History

Lesson Questions

Be sure to number your Answer Sheet 14FR-2.

Place your Student Number on every Answer Sheet.

Send in your set of answers for this lesson immediately after you finish them, as instructed in the Study Schedule. This will give you the greatest possible benefit from our speedy personal grading service.

- 1. Name the three forms of intelligence which can be sent over either wire or radio communication systems.
- 2. Why isn't sound transmitted by a perfect vacuum?
- 3. What is the speed of sound waves through air under normal conditions?
- 4. What three things can happen to sound waves which strike a material?
- 5. How can the reverberation period of a room be reduced?
- 6. What are the two important characteristics of a sound having the form of a simple sine wave?
- 7. Can the average human ear detect changes of less than 3 db in the intensity of a complex sound?
- 8. Which TWO of the following are the most objectionable types of distortion insofar as sound signals are concerned: 1, frequency distortion; 2, amplitude distortion; 3, phase distortion?
- 9. What characteristic of the human eye allows it to retain an impression of an object for a short time after that object has disappeared?
- 10. What is meant by the aspect ratio of a television picture?

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CONVERSATIONALLY SPEAKING

There is danger in thoughtless conversation. Too many of us say things, unintentionally of course, which hurt, irritate or embarrass a friend—and a word once spoken can never be unspoken. Think before you talk.

Conversation is a give-and-take proposition, and listening is the "take" part. Talk only when you can say something of interest. Otherwise, remain silent. Let your silence be eloquent enough to show that you derive pleasure from listening—that you consider the words of your companion far more valuable than anything you could say. This kind of silence can make just as many friends as good conversation.

Talk about things which will interest and please your listeners. Their hobbies, their work, their children and their homes are all good opening topics for conversation. Don't talk about yourself, your troubles or your work, unless asked.

Avoid expressing definite opinions on controversial subjects, for they often lead to unpleasant arguments. Ridicule of another person is likewise taboo at all times. If you can't say pleasant things about others, keep quiet. Finally, reserve technical discussions for technically-minded listeners.

J. E. SMITH



15FR-2

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STUDY SCHEDULE NO. 15

For each study step, read the assigned pages first at your usual speed. Reread slowly one or more times. Finish with one quick reading to fix the important facts firmly in your mind, then answer the Lesson Questions for that step. Study each other step in this same way.

The low-frequency amplifier consists of a voltage amplifier and an output stage. In this section you learn about the most popular voltage amplifier—the R-C coupled type—and learn how the fidelity of signal reproduction depends on parts values. Answer Lesson Questions 1 and 2.

5. Inverse Feedback Pages 25-30 Much of the distortion that arises in pentode and beam power output tubes can be eliminated by feeding back an out-of-phase signal so as to cancel some of the distortion. Here you learn how inverse feedback circuits of this kind work. Answer Lesson Question 8.

- 8. Mail your Answers for this Lesson to NRI for Grading.
- 9. Start Studying the Next Lesson.

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LOW-FREQUENCY AMPLIFIERS FOR SOUND AND TELEVISION RECEIVERS

Low-Frequency Voltage Amplification

R^{ADIO} signals go through several re-ceiver sections to reach the lowfrequency amplifier. As a review, you will recall that the signals (picked up by the receiving antenna) consist of radio frequency variations that follow the speech, picture, code, or music intelligence. The desired modulated carrier is selected and amplified by the r.f. and i.f. amplifiers. Next, the signal is fed into the demodulator (also called the detector). The demodulator stage eliminates the carrier, and produces a signal that is the electrical equivalent of the original intelligence-the signal can now be used to reproduce the sounds or picture we want, except that it is too weak to operate the reproducer. The job of the low-frequency amplifier is to increase the voltage or the power of this signal (from the demodulator) sufficiently to operate the reproducer.

Requirements. An increase in the signal voltage is all that is required to modulate the electron stream in the image reconstructor tube of a television set. However, the loudspeaker of a sound receiver may require considerable *power*—it is common to find sound receivers with power outputs ranging from 1 watt to as high as 10 or 12 watts. (Public address systems range up to 100 watts and more.)

By proper design, it is possible to obtain the necessary power from one or more "power" tubes, provided sufficient signal voltage can be applied to the control grid of this tube (or tubes). To obtain this signal voltage, the output of the demodulator is fed through one or more *voltage* amplifier stages. These voltage amplifier stages plus the power output stage make up the complete low-frequency amplifier. (The cathode ray tube or picture reproducer is the output stage in a television receiver.)

Since the low-frequency amplifier deals with the electrical equivalent of the original intelligence, the amplifier must of course create as little distortion as possible. Amplitude and frequency distortion must be controlled in a sound receiver, and in a television set the designer must worry about phase distortion, also.*

► In this lesson, we shall study the complete low - frequency amplifier. (Future lessons will cover r.f. and i.f. amplifiers and demodulators.) We will begin with voltage amplifiers and then go into power stages.

We find that low-frequency voltage amplifiers are classified according to the devices used to couple the stages together. Thus, we have the resistancecapacitance coupled amplifier, the impedance-coupled amplifier, and the transformer - coupled amplifier. Of these the resistance-capacitance type (commonly known as the "resistancecoupled" or as the "R-C" amplifier) is the most widely used, so we will start with it.

^{*}The low-frequency amplifier in a sound receiver is called the *audio amplifier* (because it handles frequencies corresponding to the audible range), and that in a television receiver is called the *video amplifier*.

Fig. 1 shows the details of resistance-capacitance coupling, so named because resistors and condensers serve to couple one tube to another.

You have met the amplifier before, but let's study its operation in more detail. As you have learned, the filaments are supplied with heater power, even though it is customary not to show the filament connections on schematic diagrams of the kind shown in Fig. 1. (This is purely a convenience to avoid complicating the diagram when it is being used to study the signal paths.) The plate supply is indicated by the signs B+ and B-. Grid bias is obtained from the self-bias resistors R_2 (for VT_1) and R_5 (for VT_2). The a.c. components of the plate currents are by-passed around these resistors by C_2 and C_5 respectively. That is, there is only a small a.c. voltage drop between the tube cathodes and B—, because the reactances of the condensers are lower than the resistances of the resistors they by-pass. However, the d.c. current has to flow through the bias resistors, and the resistor values are chosen so that they provide the voltage drop needed to furnish the desired grid bias.

The input signal voltage *e* causes a signal current to flow through C_1 and R_1 . The resulting voltage drop across grid resistor R_1 becomes the grid signal voltage for VT_1 . (The signal current is a.c., so it "passes through" the condenser. Hence, the entire a.c. voltage e appears across R_1 except for the portion lost in the reactance of the condenser, as we shall soon show.) Tube VT_1 amplifies the signal, so an amplified version appears across plate load resistor R_3 . This load a.c. voltage causes a signal current to pass through condensers C_3 and C_4 , and grid resistor R_4 . The resulting voltage across R_4

becomes input for VT_2 .

Condenser C_3 is called the *coupling* condenser because it "couples" the circuits of VT_1 to those of VT_2 . Condenser C_4 also could be considered as a coupling condenser, but instead it is called a by-pass condenser, because it is primarily selected to by-pass the signal so that the signal current does not flow through the B supply. (This condenser completes the a.c. circuit to ground, so that the signal is directly applied to the grounded end of resistor R_4 .) This completes the passage of the signal through the stage using VT_1 ; it now is acted upon by VT_2 .

Frequency Band. Before we can determine the gain of the stage using tube VT_1 , we must consider the range or "band" of frequencies that this amplifier is called upon to pass.

The use to which the receiver is put determines the width of the band of frequencies that the low-frequency amplifier must pass. Communications receivers, intended only for the reception of voice or code signals, have to amplify only those frequencies between about 200 and 3000 cycles.*

Sound receivers for the home must amplify a wider band for musical reproduction. Although sound frequencies range from 30 cycles to above 16,-000 cycles, the average inexpensive radio is limited to a frequency range of from about 200 cycles to 5000 or 6000 cycles. Some of the better receivers, including the average f.m. receiver, have ranges of 50 cycles to 8000 cycles. A few high-fidelity a.m. and f.m. receivers have a range of from 30 to 16,000 cycles.

In television, the desired frequency band ranges from 10 cycles to near 4,-000,000 cycles. This band is so wide

^{*}Like most radio men, we use "cycles" for "cycles per second," which is the correct term.

that special compensators are necessary, as we shall soon see.

► Even though the low-frequency amplifier is intended to pass a particular band of frequencies (as we just described), the amount that it amplifies the higher and lower frequencies in this band is different from the amplification given to frequencies in the middle of the band. That is, it is not perfect in its frequency response, and it may not be economical to improve the response beyond a predetermined level. Therefore, the parts are chosen to give a certain response, and this places limits on the replacement values that come from changes in the load, or from voltage-dividing networks that vary with frequency. Let's see what occurs in our R-C amplifier of Fig. 1.

Mid-Frequency Response. First, let's consider the response to frequencies in the middle of the band. If the amplifier is designed to pass (reasonably well) frequencies between 100 cycles and 5000 cycles, then we can call frequencies from 1000 to 3000 cycles the mid frequencies.

When we examine the parts values used in an R-C amplifier in an a.m. or f.m. receiver, we find that the capacity used as C_4 in Fig. 1 is always so large



FIG. I. This R-C coupled amplifier is the basic circuit found in most sound and television lowfrequency amplifiers.

can be used. Hence, it is advisable to learn more about how an R-C amplifier (as found in a.m. and f.m. receivers) responds to the middle, low, and high frequencies in its pass band.

FREQUENCY RESPONSE

You know that the gain of a stage depends on the amplification factor (μ) of the tube, the tube's plate resistance, and the value of the load impedance. The grid voltage is stepped up by the factor μ , and the resulting equivalent a.c. plate voltage (μe_G) is then divided between the plate resistance and the load. Therefore, since μ and the plate resistance remain the same almost regardless of frequency, any change in the over-all gain must that this condenser has negligible reactance to all frequencies in the band to be passed. Also, C_3 has little opposition to the mid frequencies, so there is practically no reactance between R_3 and R_4 at these mid frequencies. Therefore, insofar as a mid-frequency signal is concerned, R_4 is connected in parallel with R_3 and the combination forms the load for VT_1 . Thus, the equivalent circuit at the mid frequencies is as shown in Fig. 2A.

You will recall that the higher the load resistance is made, the greater the voltage gain will be. In fact, we get almost 90% of the μ of the tube as a stage gain when the load resistance is 10 times the a.c. plate resistance r_F . So you might think that we need only to increase the values of R_8 and R_4 in order to get more gain. That is true, but there are practical limits on the values we can use—we will take these up after we see what happens at the low and high frequencies.

Low-Frequency Response. When we go down to low frequencies (those lower than the mid frequencies) we find that the coupling condenser (C_3) in Fig. 1) comes back into the picture, because, as the frequency decreases. the condenser reactance increases. Hence, it begins to act with R_4 as a voltage divider. In other words, as shown by the equivalent circuit of Fig. 2B, the signal voltage developed across plate resistor R_3 is actually fed into a filter C_3 - R_4 , which passes higher frequencies better than lower frequencies. As the frequency decreases, the reactance of C_3 increases; a greater proportion of the signal voltage is then dropped across C_3 , and less appears across R_4 . Since VT_2 is operated only by the signal voltage across its grid resistor R_4 , the less this voltage is, the lower the VT_2 input signal will be. Therefore, there is less voltage fed to VT_2 , so the over-all grid-to-grid (VT_1 to VT_2) gain is less, and the gain drops still more as the frequency is decreased further. Hence, in the low-frequency range, this circuit has frequency distortion (unequal amplification of different frequencies).

High-Frequency Response. The equivalent circuit for high frequencies is given in Fig. 2C. Condenser C_3 is no bother because its reactance at high frequencies is negligible. However, we now have a new factor, the capacity C_s , which represents a number of different tube and stray circuit capacities that we have lumped together for convenience. These capacities do not appear in Fig. 1 (we will explain them shortly), but you can see their result at once. Since the reactance of C_s decreases as frequency is increased, it will act as a shunt across the resistors at higher, frequencies. In effect, this reduces the load value at higher frequencies, so more and more of the signal voltage will-be lost- $\arccos_{-r_{P}}$ within tube VT_1 . Hence we have frequency distortion at the higher frequencies also, but from another cause.

► We have given the three equivalent circuits together in one drawing so that you can compare them with each other. Notice that the schematic (Fig. 1) shows the electrical connections, but does not directly indicate how the circuit will act at different frequencies. Even the equivalent circuits of Fig. 2 indicate only the trend of operation; until parts values are known we cannot tell how serious the frequency distortion may be. As we will now show, changing the parts values will radically change the response.

EFFECT OF PARTS VALUES

We can continue to assume that C_4 in Fig. 1 is sufficiently high in capacity to have no effect on any frequencies the amplifier is designed to pass. Therefore, we can concentrate on the limits set for the parts R_3 , R_4 , C_3 , and C_8 of Figs. 1 and 2.

Value of R_3 . The value of the load is one of the factors that determine the stage gain. Since we want the stage to amplify, you might imagine that the value of plate resistor R_3 should be high. It is desirable to have this a high resistance at the middle and low frequencies, but not at the higher frequencies. This is because the higher the resistive load is, the sooner C_s becomes an effective shunt. This effect is shown by Fig. 3. Here, curve A shows the frequency response for a high load resistance, and curve B the response for a much lower load resistance (assuming the same value of C_8 in each case).

Naturally, the higher the load resistance is, the higher the amplification at the mid frequencies. Therefore, curve A shows fairly high gain at around 2000 cycles—higher than that shown by curve B.

However, compare the output at 2000 cycles with that at 5000 cycles for



FIG. 2. The R-C amplifier responds differently to the high, middle, and low frequencies in its pass band. These equivalent circuits show how the circuit components affect the response.

curve A. There is a severe loss of higher frequencies (frequency distortion) beginning at about 4000 cycles. On the other hand, curve B shows a much smaller difference between the midfrequency response and that at 5000 cycles. In other words, the gain shown by curve B drops only from 5 to about 3 in the range from 2000 to 5000 cycles, whereas the gain shown by curve A changes from 10 to about 3 over the same frequency range.

The ideal amplifier—one having no frequency distortion — would amplify *all* frequencies in its pass band equally. Hence, curve B approaches the ideal more closely than does curve A. This shows that there is good reason for keeping the effective load for the stage low when high fidelity is desired. (However, this reduces gain, and often a compromise is necessary to get reasonable gain with good fidelity.)

▶ In addition, the d.c. plate current of VT_1 passes through R_3 , and this brings up a power supply problem. The drop in R_3 (caused by the d.c. plate current) is subtracted from the B supply voltage. The remainder is the plate-cathode voltage actually supplied to the tube, and we want this voltage to be reasonably high. Since there are limits to the power supply voltage, the value of R_3 must be kept down to avoid too much d.c. voltage drop. The most common values in a.m. and f.m. home receivers range from 50,000 ohms to 250,000 ohms. Therefore, the gain of a practical R-C amplifier stage is not as high as is theoretically obtainable.

Value of R_4 . Increasing the resistance of R_4 seems to be another way of



FIG. 3. Using a higher load value gives greater gain (curve A), but there is a more rapid dropoff in the high-frequency response.

getting a higher load value, since this resistor is in parallel with R_3 , and in addition, no d.c. is supposed to flow through it. However, if the following tube $(VT_2 \text{ in Fig. } 1)$ is a power tube, we may encounter gas trouble.

No matter how well a tube is evacuated in manufacture, some gas (air) remains in it. The heat of operation may drive gas molecules out of the "pores" of the metal elements. When the gas level upsets operating conditions, we say that the tube is "gassy." This trouble occurs particularly with power tubes because of their high current levels.

When the tube becomes gassy, speeding electrons strike the gas molecules and knock out electrons, leaving positive gas ions. The positive gas ions move toward the cathode, but some "capture" electrons from the grid. This causes an external electron flow to the grid element. This electron flow must pass through R_4 , and therefore, develops a voltage across it. The grid current may be quite small at first. However, if resistor R_4 is very high in value, then even a very small gas current will cause an appreciable drop across it.

The voltage drop that occurs when the tube is gassy has a polarity that makes the grid end of R_4 positive. Hence, the drop opposes the bias voltage, and causes a reduction of the grid bias. Consequently, there is an increase in plate current. Eventually, the higher than normal plate current, and the bombardment of the cathode by the heavy gas ions will destroy the emissive properties of the cathode and thus ruin the tube.

For this reason, there is a definite upper limit to the size R_4 can have. If tube VT_2 is a power tube, R_4 must be no more than 100,000 ohms to 250,000 ohms in most cases.

If the tube is not a power tube, there is less gas trouble, and R_4 can be larger in value. You may find voltage amplifiers using values from 250,000 ohms to as high as 10 megohms.

Value of C_3 . The value of C_3 needed for good low-frequency response depends on the value of R_4 . If R_4 can be a high resistance, then even a small capacity can be used without excessive loss of the low frequencies. However, we have just shown that R_4 cannot always be made so large; therefore, C_8 may have to be fairly high in capacity.

► The, low-frequency response of -a radio receiver depends on the design of a number of things — the large amount of audio power needed for good low-frequency response requires an oversized loudspeaker, a large power supply, a high-power output stage, and a large console with a specially designed speaker compartment. All these items are costly, a fact that explains why the low-frequency response is sacrificed in the average receiver.

Once the limits have been set by the design of these components, there is no reason for making the voltage amplifier too perfect in its response; in fact, it is better not to have the low-frequency response too good, because if the amplifier can amplify very low frequencies, slight changes in the power supply voltages may cause it to oscillate. This oscillation prevents the passage of the signal, so all that can be heard is either a low-frequency howl or a "putt-putt-putt" sound called *motorboating*. Therefore, C_3 and C_1 of Fig. 1 are made no larger than necessary in practical amplifiers. Values found in the average radio are from .005 mfd. to perhaps .1 mfd., with the most common values being .01 or .05 mfd.

Value of C_s . The shunt capacity C_s is made up of the inter-electrode capacities within the tubes, and the capacities between wires and parts in the circuit.

The tube inter-electrode capacities are shown for a triode tube in Fig. 4. You will recall that a capacity exists between any two conductors separated by an insulator. Therefore, the metallic tube elements, separated by a vacuum, form tiny condensers. As shown by Fig. 5, the capacity between the grid and the cathode (called C_{GK}) is across the input circuit, and the capacity between the plate and the cathode (called $C_{\rm PK}$) is across the output circuit. Even worse is the capacity between the grid and the plate $(C_{\rm or})$. Notice that this capacity acts as a coupling condenser between $R_{\rm L}$ and the input, so that part of any a.c. voltage across $R_{\rm L}$ will also be across the input. This means that part of the amplified signal voltage across the load will feed back to the input circuit through C_{GP} . Because of the phase relationships involved, the reaction of this feedback voltage on the input is such that the circuit acts as if the input capacity of the tube were far higher than the measured capacity between the grid and the cathode elements.

Therefore, going back to Fig. 1, tube VT_2 acts as if it had a high capacity between its grid and its cathode, in parallel with R_4 . Also, tube VT_1 has an output capacity between its plate and



FIG. 4. The inter-electrode capacities within a triode tube.

its cathode (across R_3), and there is stray capacity between the chassis and the wiring and parts. (Coupling condenser C_3 is the most bulky part, so a considerable amount of stray capacity exists between it and the chassis. This is another reason for choosing C_3 with a capacity that is no larger than necessary—if its physical size is kept down, the stray capacity will likewise be minimized.)

Since the output capacity of VT_1 and the input capacity of VT_2 are in parallel, they can be lumped together and considered as a single condenser. Therefore, at high frequencies, the circuit is equivalent to that shown in Fig. 2C, where capacity C_8 represents all the shunting capacities.

 \blacktriangleright Set designers make every effort to minimize the value of C_8 . Once this



FIG. 5. The plate-grid inter-electrode capacity provides a coupling path so that the voltage across the load is transferred to the input circuit.

value is determined, it fixes the largest values of R_3 and R_4 that can be used and still have the required fidelity.

Summary. Tube VT_1 is usually chosen to be a high-mu triode (or pentode in some cases). Tubes with high gain and low d.c. current values have been specially designed for use in R-C amplifiers. Even so, fidelity compromises mean that R-C amplifiers are used where gain is less important than fidelity, because it is possible to get good fidelity, despite what has just been shown, if the gain can be sacrificed. Hence, we can make the following rules for circuits like that in Fig. 1:

1. For maximum gain at the middle and low frequencies, R_3 and R_4 should be high in resistance.

2. For minimum loss of high frequencies, R_3 should be low in resistance.

3. The larger C_3 is in capacity, the better is the low-frequency response. As a serviceman, you will have to replace these parts whenever defects occur in them. You should use replacements having values within 20% of the original values if you do not wish to change the circuit characteristics. If you do want to make a change, or if you do not know the original values, the rules just given will help you choose the best replacements for the desired conditions. When you install replacements, take care to minimize the shunting capacity C_s . Don't add more wire than necessary, and don't disturb positions of parts more than is necessary.

More About Voltage Amplifiers

So far, we have considered the frequency response and frequency distortion of a resistance-coupled amplifier. Equally important is the amplitude distortion that can occur—in fact, even small amounts of amplitude distortion can be far more objectionable than fairly large amounts of frequency distortion. Let's continue with the resistance-coupled voltage amplifier as our example, and learn more about amplitude distortion.

AMPLITUDE DISTORTION IN R-C AMPLIFIERS

You know that amplitude distortion is produced whenever the wave shape of a signal is changed. This distortion will occur whenever the tube's operating characteristic is curved. Therefore, to minimize the distortion, operation is limited to the straightest portion of a tube's characteristic.

► However, so far we have presented the darker side of this story because we have shown you only the characteristic curve of the tube *alone*. Such "static" characteristic curves show how the tube operates when there is *no plate load*. These curves are the kind the tube manufacturer can most easily furnish since he doesn't know just what load values may be chosen by the set designer. However, when the tube is operated with a load (the normal condition) its operation is over a "dynamic" characteristic that is considerably different in shape. Naturally, we are interested in dynamic curves because they show how a tube actually operates in a stage.

Dynamic curves can be found by actually setting up the circuit and measuring the response, or they can be plotted from the static curves. We won't go into their construction here, but the following facts will show why they differ so much from the static curves.

First, you know that the plate current of a tube with no load increases rapidly as the grid bias is made less negative. Curve A of Fig. 6 is a typical static curve, showing how the plate current and the grid voltage are related when the tube plate voltage is a fixed amount.

Next, you have learned that when a load resistor is used, some of the B supply voltage is dropped across it because the tube plate current flows through this resistor. Therefore, the actual plate-cathode voltage of a tube is the difference between the d.c. load drop and the supply voltage.

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 \blacktriangleright Now, let's suppose we change the bias so that the grid is made less negative. Since the plate current tends to increase when the bias is less negative, there will be a greater drop across the load resistor, so the actual tube plate voltage drops. This reduction in plate



FIG. 6. Curves B and C show how increasing the load resistance tends to flatten the tube E_G-I_P characteristic curve.

voltage counteracts some of the plate current change (the plate current depends on both the grid voltage and the plate voltage) so the actual change in current for a particular grid voltage change is not as large when a load is used.

Hence, when a load is used, the tube characteristic is "ironed out" so that it is much flatter and straighter than the static curve. The effect of this selfregulation is shown in Fig. 6, where curves B and C represent dynamic curves for two different load values for a triode tube.

You can see that there is far less curvature in these dynamic curves than there is in curve A. This means that there will be much less amplitude distortion when a load is used. That is, a change in grid voltage (such as is caused by a signal) will cause much more nearly linear plate current changes

Furthermore, the higher the load resistance is, the flatter the curve. However, this brings back our compromise conditions; you know that high resistances cause excessive loss of the higher frequencies. Therefore, the designer must consider gain, frequency distortion, and amplitude distortion when he chooses values for the tube load.

CASCADE AMPLIFIERS

These factors are further complicated when there is more than one stage of R-C amplification. For example, suppose we have two stages, as shown in Fig. 7. We'll consider tubes VT_1 and VT_2 to be the voltage amplifying tubes for these stages, with VT_3 shown only to indicate the output coupling. Furthermore, let's assume that the characteristics of corresponding parts are alike: that is, the tubes are the same, condensers C_4 and C_7 have the same capacity, and resistors R_3 and R_6 have the same resistance, as do R_4 and R_7 .

The first thing we are interested in is the amount of amplification we get. Let's suppose that each of the stages is capable of giving a gain of 10 at the mid frequencies, and that a signal of 1 volt is applied to the input of VT_1 . There will be a signal of 10 volts across R_3 ; this voltage is applied to tube VT_2 .



FIG. 7. A two-stage R-C amplifier.

World Radio History
This tube also has a gain of 10, so its 10-volt grid signal is raised to 100 volts^{*} across R_6 . Thus, the gain of successive stages is the *product* of their respective gains; the gain of the VT_1 stage is multiplied by the gain of the VT_2 stage (10 x 10 — 100 and our 1volt signal was raised to 100 volts).

When more than one stage is used, we say that the stages are "in cascade," because each one steps up the signal output of the previous ones by its own gain factor.

Fidelity. From this, we might at once assume that all we need to do to with two stages it will be one-quarter (one-half times one-half) that of the mid frequencies.

► This is one of the reasons why the number of low-frequency amplifier stages is kept at a minimum in radio receivers. It is far better to use a single high-mu triode or pentode tube than to use two or more stages of low-gain triode tubes.

TELEVISION AMPLIFIERS

Although careful design is necessary to have high fidelity in a resistance-



FIG. 8. Curve A shows that there is a higher midfrequency gain when two stages are used, but there is a more rapid drop-off at the lower and higher frequencies.

obtain more gain is to add more stages. However, if we do, we find that the fidelity becomes much worse when more than a single stage is used.

For example, see Fig. 8. Curve B^{-1} represents the fidelity of a single stage, and curve A represents that of two identical stages. Curve A shows that the gain is increased at the mid frequencies, but you can see that the addition of another stage exaggerates the frequency distortion already present at the low and high frequencies in one stage. Thus, if the gain at some particular frequency is only half that of the mid frequencies in a single stage,

*If the input was .01 volt, the output of VT_1 would be .1 volt, and the output of VT_2 would be 1.0 volt.

coupled amplifier, this circuit still offers a greater range of frequency response than any other, so it is universally used (in a modified form) as the video intelligence signal amplifier in television receivers. We will leave complete details for later lessons, but some of the methods used to get a broad frequency pass band are of particular interest at this time.

For television, the frequency range is from about 10 cycles to as high as 3,000,000 or 4,000,000 cycles. This is many times greater than that found in any sound receiver. Getting this extended high-frequency response is quite a problem.

To reduce the effect of the shunting capacity (corresponding to $C_{\mathbf{s}}$ in Fig.

2C) and thus obtain better high-frequency response, a very small load resistor (1000 to 5000 ohms) is used. This means extremely low gain unless special high mutual-conductance pentode tubes are used.

Other steps are also taken to reduce the effects of the shunt capacity. As you learned, the shunt capacity is due



FIG. 9. The effects of the shunt capacities $C_{\tt A}$ and $C_{\tt B}$ are reduced by using $L_1.$

partly to the output capacity of one tube and partly to the input capacity of the other. In Fig. 9, these capacities are represented as C_A and C_B . One scheme used is to insert a coil (coil L_1 in Fig. 9) between the plate circuit of VT_1 and the grid circuit of VT_2 . This serves to isolate C_B from R_2 , so it reduces the effects of C_B on the load of VT_1 . The isolation increases with frequency increases because the reactance of L_1 increases with frequency. Hence, its increasing reactance counteracts the decreases of the C_B reactance insofar as VT_1 is concerned.

Also it is possible to make L_1 and C_B resonant at some rather high frequency. This forms a series resonant circuit so that a maximum voltage appears across C_B at the resonant frequency. This voltage is applied to VT_2 , so the response at and near this resonant frequency is improved.

Another scheme is shown in Fig. 10. Here, coil L_2 is placed in series with R_2 . As the frequency increases, so does the reactance of L_2 . Therefore, as L_2 is part of the load, the load impedance rises as the frequency increases. Al-

though the shunting effects of the stray capacity are increased, there is a better division of the μe_{α} voltage between the $r_{\rm P}$ of VT_1 and the load, so the VT_1 stage provides higher gain at the higher frequencies. This tends to minimize the drop-off in high-frequency response. Furthermore, by making L_2 resonant with C_{A} (at a high frequency), a parallel resonant circuit is formed. This provides a rise in the load resistance at resonance, which in turn further boosts the gain at the high frequencies. Sometimes both schemes shown in Figs. 9 and 10 are used in the same amplifier. ▶ In addition to these methods of reducing shunt capacity effects, television engineers use sockets that have low capacity to chassis, and mount parts so that they are separated from the chassis as much as possible, yet try to keep the leads between stages short to avoid capacity between these leads and other wires. These precautions are necessary because of the very wide



FIG. 10. Another compensation for shunt capacity is to use L_2 in series with the load resistance.

band of frequencies passed by the television amplifier. Naturally, it is not necessary to go to such extremes in ordinary sound receivers.

USING COILS FOR COUPLING

The resistance coupling you have just studied is the coupling method most commonly used today for voltage amplifiers. However, some amplifiers



FIG. 11. Three forms of impedance coupling are used in low-frequency amplifiers.

use inductances (coils) for coupling. Some of these inductances are choke coils, others are transformers—we will study both.

Impedance Coupling. Essentially, an impedance-coupled amplifier is like a resistance-coupled amplifier except for the substitution of one or more iron-core choke coils for the resistors. Fig. 11A shows a full impedance- coupled amplifier. Here choke coil L_1 is used as the plate coil, and choke coil L_2 is used as a grid coil for tube VT_2 . The operation is similar to that of resistance coupling except for the effects of the coils on the frequency response, and the effect of L_1 on the plate voltage.

Since L_1 has a low d.c. resistance, there will be very little d.c. voltage throp across it. Practically the full supply voltage is applied to the plate of the tube. For this reason, this circuit is particularly useful where the B supply is very limited, as in batterypowered circuits.

Coil L_2 also has a low d.c. resistance. Therefore, any gas current flowing through L_2 will cause but a small voltage drop, so the use of a coil eliminates the bias shift encountered when a grid resistor is used with a gassy tube.

► Since both choke coils offer higher reactances as the frequency increases, we might expect the gain of the stage to go up as the frequency rises. However, good high-frequency response is not obtained, because the choke coils have rather high distributed capacities, which are effectively in parallel with the coils. The distributed capacity tends to cut off the high-frequency response even more rapidly than in a resistance-coupled amplifier, because the coil capacities are added to the normal shunting capacities in the circuit.

The low-frequency response is poor also, because the reactance values drop at low frequencies. Hence, unless very large values of inductance are used, the low-frequency response will be much poorer than that of the mid frequencies.

Fig. 11B shows a variation of this circuit in which coil L_2 is replaced by resistor R_2 . This eliminates the distributed capacity and inductance troubles of one of the choke coils, so this circuit is somewhat better in frequency response than that shown in Fig. 11A.

This is the impedance-coupled circuit most likely to be found in use, for it is considerably cheaper than the one in Fig. 11A.

Fig. 11C shows another possible variation, in which coil L_1 is replaced by resistor R_1 . (Choke L_2 is used as the grid coil.) This circuit has little to recommend it unless tube VT_2 is likely to be gassy and a high resistance is not wanted in the grid circuit.

Transformer Coupling. Fig. 12 shows an example of a transformercoupled voltage-amplifier stage. As you learned in an earlier lesson, transformers have a special kind of amplification also; they can step up the signal voltage by their turns ratio.

Therefore, if tube VT_1 in Fig. 12 has a gain of 10, and the transformer T_2 a step-up ratio of 3, the first stage gain will be the product of the two, or 30. Hence, transformer-coupled amplifiers give much higher gain for the same type of tube than do either impedance-coupled or resistance-coupled amplifiers.

As you learned in an earlier lesson, the frequency response of a transform-



FIG. 12. A typical transformer-coupled, lowfrequency, amplifier stage.

er is limited. The inductance of the primary winding must be very high to prevent an excessive loss of low frequencies because of the voltage division betwen the primary and the plate resistance of tube VT_1 . In addition, the distributed capacity of the windings must be low to prevent an excessive loss of the high frequencies. Essentially, transformers give a frequency response similar to that obtained with impedance coupling.

► The resistance-coupled amplifier offers far better fidelity than either of the devices using iron-core coils. However, the high amplification obtainable from the transformer makes it a favorite where gain is more important than frequency response.

Single-Ended Power Amplifiers

Once the voltage amplifier has made the signal sufficiently large, the voltage is fed into a power stage which supplies the power to drive the reproducer (a loudspeaker in a sound receiver). On paper, the power stage diagram is very similar to that of the other audio stages. It is not until we notice the plate current values and bias voltages that we see the difference.

Modern receiver loudspeakers require a considerable amount of a.f. power—anywhere from 1 watt to 12 watts on maximum output levels. Ordinary voltage amplifying tubes cannot furnish anywhere near this power. Most of them draw only from .25 watt to about 1 watt of d.c. power. This power is found by multiplying the B supply voltage by the d.c. plate current. For example, if the B supply voltage is 250 volts, and the d.c. plate current is 1 ma. (.001 amp.), then the d.c. power is 250 x .001, or .25 watt.

The total developed a.f. power is always far less than the d.c. power, and at least half the a.f. power is lost in the plate resistance of the tube. Therefore, we must use an output tube capable of delivering a higher power than can a voltage amplifier.

This means a power tube must draw a higher d.c. power than a voltage amplifier in order to deliver the required a.c. power. Since it is desirable to operate all the stages at the same plate voltage to simplify power pack



FIG. 13. The voice coil of the loudspeaker is matched to the plate impedance of the output tube by the output transformer.

design, power tubes are designed to draw more plate current. (Power is equal to voltage times curent, so increasing the current increases the power.)

▶ To get the increased emission, the size (length and diameter) of the cathode is increased, and the filament current is increased to the value needed to heat the larger cathode. Then, the plate area is increased to correspond to the increased cathode size. This provides more electron paths (in parallel) from the electron cloud to the plate, and, as the same plate voltage acts across each, this has the effect of resistance in parallel—there is less plate-cathode resistance, so the plate current is higher for the same plate voltage.

For comparison, audio voltage amplifying tubes in radio receivers draw currents of from .5 ma. to perhaps 5 ma., but power output tubes draw currents of 30 to 50 ma. and more.

Coupling to Speaker. Fig. 13 shows the method of coupling from the plate of a power tube to the voice coil of a dynamic loudspeaker. Usually a voice coil has an impedance of only 4 to 8 ohms, which is far lower than the plate resistance of the tube. Transformer T_1 is used to match the two impedances.

Load Matching. As you know, maximum power output is obtained when the load impedance exactly matches the plate resistance of the tube. This does not assure minimum amplitude distortion. However, you will recall that increasing the load re-



Magnetic speakers were once very popular, and in radio servicing you may still encounter a receiver using one. An output transformer is not needed with them—they are made with many turns of wire, so it is possible for them to have impedances of 2000 to 6000 ohms, which is a close match to the required load for output tubes. In the direct coupling shown at A, the d.c. plate current flows through the speaker. This may be undesirable, for, if the wrong connections are made, the resulting d.c. flux will be opposite to that of the permanent magnet used in the speaker. This may ruin the speaker magnet. Also, the high current requires large wire sizes, which may make it difficult to wind the coil in the space provided. Hence, many sets used the choke-condenser coupling shown at B to avoid this; here the d.c. is blocked by C₃ and flows only through L₁.

sistance value tends to flatten the triode tube characteristic curve. Therefore, it has been found that with *triode* tubes, a load equal to about twice the plate resistance value will give the maximum undistorted power output.

Notice the difference. The maximum power, with some distortion, is obtained when the load and source values are matched. The maximum *undistorted* power (less than the maximum power but with less distortion) is obtained when the load is twice the tube resistance. (Further increasing the load resistance reduces the distortion some more, but the power output falls off because of the mismatched impedances.)

The load values and operating voltages listed in tube charts for triode power tubes are for the *maximum undistorted* power output, and the output transformer is designed for these values.

PENTODE AND BEAM POWER-OUTPUT TUBES

The triode power-output tube offers very good fidelity, but it requires a high signal input to be driven to its full output. Also, it consumes a considerable amount of d.c. power. The latter is no problem in a large receiver with a heavy-duty pack, but is troublesome in a set using battery or a.c.d.c. supplies, which are limited in their power. For these reasons, pentode and beam power tubes having higher efficiencies than triodes* were developed.

The pentode and beam tube types require considerably less grid signal voltage than does a triode to produce the same output. For example, the type 2A3 triode tube requires a grid signal having a peak value of nearly 45 volts to give 3.5 watts output. On the other hand, the 6K6G pentode power tube requires a signal of only about 18 volts peak to give 3.5 watts output.

This is important because it leads to a simpler voltage-amplifier design. In most receivers, one voltage-amplifier stage is plenty with pentode or beam output tubes, but triode output tubes frequently require two.

The Pentode Tube. Fig. 14 shows an output stage using a typical power pentode tube. As you will notice, this tube has three grids. From the cathode toward the plate, these grids are, respectively, the control grid, the screen grid, and the suppressor grid.

The control grid performs the same



FIG. 14. The elements of a pentode power output tube are identified.

function as it does in a triode tube. The screen grid is a relatively coarse grid that is connected to a positive voltage—the screens of power types are usually connected directly to B+. Its high potential causes it to draw electrons from the electron cloud. The screen grid catches some of these, but because of its coarse structure (widely separated turns of wire) most of the electrons go through the mesh to the plate.

Since the screen grid is considerably closer to the cathode than the plate is, the plate current is much more dependent on the screen grid voltage than it is on the plate voltage. If the screen grid voltage is kept constant, the plate voltage can be varied over rather wide

^{*} For comparison, the a.c. power output (with permissible distortion levels in the case of the pentode) of the 6K6G pentode and the 2A3 triode are similar (about 3.5 watts). With the same plate voltage, the 2A3 draws 60 ma., but the 6K6G draws only 32 ma. The d.c. powers used, at 250 volts, are 15 watts for the 2A3, and 8 watts for the pentode.

limits without changing the plate current to any considerable extent.

Hence, in pentodes, the control grid has far more effect on the plate current than does the plate voltage, so the pentode has a high amplification factor. (As you know, the amplification factor or μ of a tube is the ratio of the plate voltage change to the grid voltage change that produces the same plate current change.) This means that to obtain the same a.c. plate voltage (μe_G) and hence the same a.c. power, the grid signal voltage required by a pentode is much smaller than that required by a triode power output tube. Thus, the power sensitivity of a pentode is much higher than that of a triode.

► The screen grid gives us the desired high sensitivity, but the presence of this second positive element causes a trouble that requires the presence of the suppressor grid.

The screen grid attracts electrons and speeds them on their way to the plate (in fact, it is sometimes called an accelerating grid). When these high-speed electrons strike the plate, other electrons are knocked out of the plate material. This emission of electrons due to bombardment is known as secondary emission.



FIG. 15. The internal construction of the beam power output tube.

As long as the plate is more positive than the screen grid, these secondary emission electrons are attracted right back, so no harm results. However, the voltage developed across the plate load varies with_the signal, and, of course, this drop will at times be so large that the plate voltage (supply minus load voltage) falls below the screen grid value. Whenever this occurs, the secondary emission electrons tend to move from the plate to the more positive screen grid. This reversal of electron flow will reduce the plate current, causing distortion.

Secondary emission is not prevented by the suppressor grid, but its bad effects are removed. This grid, tied to the cathode, is always at the cathode potential. It is therefore always negative with respect to the plate, and repels any electrons that try to move away from the plate, driving them back. Only the slower-moving secondary emission electrons are controlled by the suppressor grid; the high-speed electrons moving from the electron cloud fly past this grid so fast that its potential cannot block them.

Beam Power Tubes. The beam power tube is practically a pentode tube. However, instead of a suppressor grid, it uses a pair of beam-forming electrodes to get the same effect. Fig. 15 shows a cut-away view of such a tube.

The beam-forming electrodes are connected to the cathode, so they are negative with respect to the plate. They repel electrons, forcing the electrons to "bunch together" and pass through the openings between these electrodes to reach the plate. Thus, they force the electrons to form two streams or "beams." This concentrates the flow of electrons in a stream, actually producing a compact mass of them between the plate and the screen grid. Thercfore, any secondary emission electrons trying to move from the plate toward the screen grid are met at once by a large body of negative electrons (practically an electron cloud between the plate and the screen grid) that forces them back to the plate. resistance of the 6K6G is about 50,000 ohms). Notice, too, that increasing the load resistance does not improve the shape of the characteristic. In fact, the opposite occurs; an increased load resistance increases the curvature at the right end of the dynamic curve.



FIG. 16. These curves are typical of the static and dynamic curves of pentode power output tubes. Notice the upper bend curvature increases with load increases. Hence, the load must be kept low to keep down the third-harmonic distortion.

PENTODE AND BEAM POWER DYNAMIC CHARACTERISTICS

The dynamic characteristics of pentode and beam power tubes are not as linear as are those of a triode. Fig. 16 shows the static curve for a typical pentode output power tube and also the dynamic curves for several load values.

Notice that the load values given in Fig. 16 are far below the plate resistance for the tube (the a.c. plate This undesirable drop-off in the curve occurs because the plate voltage falls below a critical value. That is, the plate current of a pentode or beam power tube is relatively independent of the plate voltage until the load voltage drop causes the plate-cathode voltage to fall about 75 volts. Further reduction of the plate voltage causes the plate current to drop, too. Therefore, the plate current follows the grid voltage changes only up to this point where the plate voltage "takes over."

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FIG. 17. Parallel output tubes were once commonly used to get an increased power output. However, the parallel arrangement is not ideal; it is subject to oscillation whenever the two tubes are not EXACTLY alike in their characteristics. (Aging will change them even if they were alike when new.) The feedback occurs through the grid-plate capacity of the tubes; to absorb this power, resistors R1 and R2 are sometimes added in series with the grids of the tubes as shown. These resistors do not have any current flow through them unless there is feedback current through the tubes. Any such feedback current must flow through the resistors, causing a loss of feedback energy. Sometimes resistors are necessary in the plate circuits as well, to prevent os-

cillation in the higher powered circuits.

The curves shown in Fig. 16 indicate that increasing the load does not remove all the second harmonic distortion, and actually increases the third harmonic distortion. (As you learned in an earlier lesson, second harmonics are added by operation over a single curvature, and third harmonics result if there is a double curvature in the tube characteristic.) Since third harmonic distortion is very distressing to the human ear, it is impossible to use a load value anywhere near the plate resistance value for this tube. Actually, the recommended load value is usually about one-seventh to one-ninth the plate resistance value for pentodes and beam power tubes.

The 6K6G tube, used with the recommended load value of 7600 ohms, will give us considerable power output, but the output will have a fairly high harmonic content when compared to that of a triode. Thus, a pentode will give us a greater power sensitivity and greater efficiency than a triode, but only at a cost of increased distortion.

Beam power tubes are not quite as bad in this respect as are pentodes, but the triode can still give far better fidelity than either. However, the great power sensitivity of the pentode and beam tubes has made them widely used, and some of the limitations on their fidelity have been removed by a special circuit known as inverse feedback. We'll study this later in the Lesson.

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PREVENTING OSCILLATION

You may have wondered at the use of C_3 in Fig. 14. In most circuits we try to *avoid* plate-to-cathode capacity, since it reduces the high-frequency response. However, particularly in power stages using pentode and beam power tubes, C_3 is necessary to prevent oscillation.

: The power output stage has rather high load voltages, so even a small percentage of feedback to the grid circuit may supply enough power to allow the stage to oscillate. Furthermore, the leakage inductance of the output transformer makes the plate load inductive. This means the load voltage is in the proper phase to cause oscillation if it is fed back to the grid circuit.

The frequency of the oscillation may be well above the audible range because resonant circuits are formed by the long leads and the stray capacities in the stage. However, regardless of the frequency of oscillation, the circuit will be blocked and signals cannot pass. C_3 is an effective by-pass at these high frequencies and reduces the feedback enough to prevent oscillations.

Since the larger the capacity of C_3 is, the more it will by-pass the higher audio frequencies, the smallest value that will adequately prevent oscillation is used. Commonly used values range from .001 mfd. to .02 mfd.

Usually, the set is designed so that C_3 and the output transformer resonate near the high-frequency end of the pass band and thus prevent some of the drop-off at these frequencies.

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In some receivers, C_3 may be connected directly across the primary of T_1 , or else its lower end may go to the chassis instead of to the cathode. These changes do not affect the results as long as C_2 and C_1 are effective.

PARALLEL OUTPUT TUBES

When more power output is needed than can be obtained from a single tube (and a more powerful tube is not available or is undesirable because of voltage requirements), more than one output tube must be used. One way in which two or more tubes can be used is to connect them in parallel, as shown in Fig. 17. The corresponding partsplates, grids, and cathodes—of the two tubes are tied together.

Since the plate resistance of the two tubes in parallel is only half that of

one tube alone, the load used is likewise only half the value that would be used with a single tube. Both tubes furnish current, so the load current is doubled. The result is that the power output of this parallel arrangement is twice that of a single tube for the same applied signal voltage.*

However, the parallel output stage is unstable (oscillations occur easily), and the fidelity is no better than that of a single output tube. When another arrangement, known as the push-pull connection, was developed, the parallel circuit was abandoned for receivers. The push-pull circuit provides even more power from two tubes than does a parallel connection and gives better fidelity. We'll study this circuit next. Incidentally, to distinguish between the push-pull and other connections, any power stage using a single tube or parallel tubes is known as a singleended stage. The push-pull stage is called a double-ended stage.

* The power is I²R, so doubling the current would give four times the power, but halving the load resistance gives us a net of twice the output.

Push-Pull Power Stages

The maximum power output that can be usefully obtained from a tube depends on the amount of distortion permissible. The push-pull circuit is particularly desirable because it eliminates all even harmonics that result from amplitude distortion in this stage. Thus, it has far better fidelity than does the single-ended output stage. This leaves only the odd harmonics so the power output can be increased until the odd-harmonic distortion level becomes objectionable. In fact, it is possible with the push-pull circuit to obtain considerably more than twice the undistorted power that a single tube could deliver.

Essentially, a push-pull circuit is one in which the grids of two tubes are fed out of phase, and the resulting plate currents are recombined so that any even (second, fourth, etc.) harmonics that are produced by distortion in the stage are cancelled. (It is important to realize that any harmonics that are part of the signal will pass through the stage; it is only harmonics added within the stage that are cancelled.) This circuit is used with triode, pentode, and beam power tubes; we'll take up each in order.

TRIODE PUSH-PULL STAGES

The basic triode push-pull circuitis shown in Fig. 18. Tubes VT_1 and VT_2 are identical types. The plate sup-(ply is fed through the center tap of transformer T_2 , so that the plate current of VT_1 flows through one half of the transformer while the plate current of VT_2 flows through the other half. Bias is obtained as a result of the flow of the combined plate currents through R_1 from B— to the cathodes.

The grid circuits are completed through the secondary of transformer T_1 to its center tap, where the return circuit goes through ground to R_1 .

Notice the direction of the electron flow in each plate circuit, as indicated by i_1 and i_2 . In each tube, electrons move from cathode to plate. Therefore, the two d.c. plate currents flow in opposite directions through halves of transformer T_2 . Assuming we have identical tubes with the same bias and no signal, these plate currents are practically equal. Therefore, since the plate currents are equal in value, and flow in opposite directions through T_2 , the fluxes produced by these d.c. currents cancel each other, and no resulting flux exists in the transformer core when there is no signal. (When current flows one way through a transformer, the flux lines in the core have a particular direction. When the current is reversed, the flux direction reverses. Therefore, if equal currents flow through equal windings in opposite directions, the two fluxes cancel each other.)

Tracing The Signal. Now let's assume that a sine-wave signal e is applied to the primary of transformer T_1 in Fig. 18. A larger voltage will ex-

ist across the secondary, since this is . a step-up transformer.

The induced voltage exists across the entire secondary winding of $T_{1.}$, Therefore, when terminal 1 of the secondary of T_1 is positive, terminal 3 must be negative, and vice versa.

For the moment, let's assume that terminal 1 is positive. Then, the voltage e_1 betwen terminals 1 and 2 is that applied to VT_1 . This is half the total ' voltage, and at the moment, the grid end of the winding is positive. Therefore, at this instant, the positive signal ' voltage makes the grid of VT_1 less negative (it subtracts from the bias voltage), so the plate current of VT_1 increases.

At the same instant, terminal 3 is negative with respect to teminal 1. In fact, it is the most negative point on the winding, so the voltage e_2 (between 2 and 3) is negative. This signal voltage adds to the bias, making the grid of VT_2 more negative, so this tube's plate current decreases. Thus, you can see that the two grids are fed signals that are 180° out of phase, because when one is positive the other is negative.

Since on this half-cycle the current



FIG. 18. A typical triode push-pull stage. In some cases you may find that by-pass condensers C_1 , C_3 , and C_4 may not be used in push-pull stages of this type. Condenser C_1 is not required if the tubes are balanced (as you will see when you study inverse feedback), and C_3 and C_4 are needed more with beam power tubes than with triodes.

 i_1 of tube VT_1 is increased, and at the same time the current i_2 of VT_2 is decreased, we get an important action in the transformer T_2 . This changing plate current has an a.c. component, so an a.c. voltage is produced across the primary of T_2 . Let us say that terminal 4 is becoming more positive than 5, while 6 is becoming more negative than 5. Hence, terminals 4 and 6



FIG. 19. When the operating point is on the lower bend of the tube's characteristic, the plate current pulses i₁ and i₂ will be distorted.

are opposite in polarity. Insofar as the entire primary is concerned, therefore, the two voltages add together and act like a single a.c. voltage of twice the value of either component. Also, the resulting a.c. current in the primary causes an a.c. flux in the core, which in turn induces a voltage in the secondary of T_2 .

▶ In the next half-cycle the polarities of e_1 and e_2 are interchanged (but they stay 180° out of phase) so the two actions reverse: current i_1 now decreases and i_2 increases. This reverses the phase of the a.c. voltage across the primary of T_2 and so reverses the direction of the flux in the core.

Since the a.c. voltage across the primary of T_2 is produced by *two* tubes, it is twice the value produced by one tube; in other words, we have double the power output of a single tube.

However, it is possible to get more from this stage, because distortion is reduced to the point that a greater output is obtainable. Let's see how.

Distortion Cancellation. Let's again apply a sine-wave signal and operate near the lower bend of each tube's characteristic curve, as shown in Fig. 19. Considering first the grid signal e_1 shown by the solid-line curve, the corresponding plate current curve is i_1 —also shown by a solid curve. The plate-current wave shape is not equal in amplitude on the two sides of its zero axis—that is, the alternation M-N is greater than the alternation M-O—so we have added the second and other even harmonics to the original fundamental sine wave.

At the same moment the other tube is getting the grid voltage e_2 . Therefore, as shown by the dotted lines in Fig. 19, its plate current i_2 is distorted also on the *M-O* alternation. However, since e_1 and e_2 are 180° out of phase, the plate current alternation of i_1 that is distorted occurs at the same moment as the portion of i_2 that is not distorted (and vice versa).

▶ Now let's examine the action occurring in T_2 when these plate currents flow in the primary. As shown in Figs. 19 and 20, the plate currents i_1 and i_2 are 180° out of phase. However, remember that they flow in opposite directions through the halves of the primary of T_2 . Therefore, if i_1 produces flux f_1 as shown in Fig. 20, then i_2 will produce the flux f_2 . In effect, the reversal of plate current direction through the primary has resulted in f_2 being "flipped over." And, as we explained earlier, the two fluxes add together as if they were produced by a single a.c. voltage across the entire primary winding. Hence, when f_1 and f_2 in Fig. 20 are added (thus, A-B plus C-D equals E-F), the resultant flux

 f_c is NOT AS DISTORTED as are f_1 and f_2 . The second-harmonic distortion present in the plate currents has been eliminated by the action of the fluxes in T_2 .

► Also, the same cancellation will occur on any other *even* harmonic, so no even-harmonic distortion of any consequence is produced in a push-pull stage. (IMPORTANT: Harmonics in



FIG. 20. The even-harmonic distortion cancellation occurs because of the manner in which the fluxes add in the output transformer.

the original signal are not cancelled, because they act like any original signal applied to the push-pull stage. This action occurs only for harmonics added by the push-pull tubes.)

Third Harmonics. Third harmonics are not cancelled, because odd-harmonic distortion affects both halves of the flux cycle, and adding the two flux components would not remove the distortion. This is shown in Fig. 21. Flux f_1 has a square-top shape (indicating odd harmonics have been added), and f_2 has exactly the same shape. Adding the two together *reinforces* the odd harmonics instead of producing cancellation. The resultant f_c thus is just an enlarged version of its two components and contains just as many third harmonics (or other odd harmonics).

Summary. We can then say about push-pull circuits:

1. The d.c. plate currents are equal and opposite, so the d.c. fluxes in T_2 cancel.

2. The a.c. plate currents are 180° out of phase, but since the currents flow in opposite directions through T_2 , the fluxes add in the core of the transformer.

3. If even harmonics are added by amplitude distortion in the push-pull stage, they are wiped out when the fluxes are added in the output transformer.

4. If odd-harmonic distortion occurs in a push-pull stage, no cancellation exists—odd harmonics go through the



Perhaps the second-harmonic cancellation can be seen easier by considering the components of the fluxes. Flux f_1 is the result of adding a fundamental A to a second harmonic B that is 90° out of phase with it. Similarly, f_2 is the result of combining the fundamental C with the 90° out-of-phase second harmonic D. With f_1 and f_2 in the relationships shown, you can see that the fundamentals A and C are in phase and can be added directly. However, the second harmonics B and D are 180° out of phase with each other, so they cancel each other. circuit in the same way as the fundamental wave.

► These facts are true whether the even and odd harmonics are the second and third or are the sixth and seventh —all even harmonics are eliminated, and odd harmonics are passed on.

Triode tubes have considerably more *even*-harmonic distortion than



FIG. 21. Odd-harmonic distortion is not cancelled in push-pull circuits because both halves of the flux cycle are distorted, so when they are added, the combination must have the same distortion.

odd-harmonic distortion. Therefore, the push-pull circuit arrangement with triode tubes in a class A amplifier comes very close to the ideal in giving distortionless output.

Increasing the Output. The elimination of even-harmonic distortion in a push-pull stage permits us to get an undistorted output from the stage, that is more than twice as large as the undistorted output a single tube can give. We can get this greater output by doing two things:

1. We can make the load more nearly equal to the plate resistance.

2. We can adjust the bias so that the operating point is nearer the lower curved section of the characteristic, thus permitting a larger grid signal voltage, which in turn means a greater plate-current swing and more a.c. power, Of course we could take both these steps with a single-tube amplifier, but doing so would increase the even-harmonic distortion in the amplifier. Such increased distortion also occurs in the push-pull stage, but the action of the stage promptly wipes it out; as a result, we get increased power without increased even-harmonic distortion. We can get about 2.5 times as much output power from a push-pull stage as from a single-ended stage before odd-harmonic distortion begins to become objectionable.

PENTODE AND BEAM POWER PUSH-PULL STAGES

Pentode and beam power tubes are used in push-pull arrangements in the same way that the triode is. As shown in Fig. 22, the only difference is the substitution of a pentode (or a beam



FIG. 22. A pentode push-pull power output stage.

power) tube in place of each triode. The operation of the stage is essentially the same.

However, pentode and beam power tubes have greater amounts of distortion, particularly third-harmonic distortion. Therefore, with push-pull stages using these tubes, it is common practice to use a low-resistance load.

As was shown for the 6K6G tube in Fig. 16, a small load causes less oddharmonic distortion because it causes less of a bend in the dynamic curve at the right-hand end. Such a load may cause more bend in the lower part of the characteristic, and thus cause more even-harmonic distortion—but we don't care; the even-harmonic distortion is eliminated by the push-pull arrangement.

In practice, a load resistance value is chosen to give all the power possible at the permissible value of odd-harmonic distortion. If, say, 5% third harmonic is permissible, the load resistance value is increased (thus increasing the power output) until the third-harmonic distortion reaches this value.

CLASS B PUSH-PULL STAGES

The circuits we have just discussed have been adjusted for class A operation. The operating point is chosen so that the grid will never be swung positive, nor will the plate current ever be zero.

However, where high power and high efficiency are required, a push-pull circuit can be operated as a class B amplifier (that is, operated with a bias set at the plate current cut-off point) without creating excessive distortion. As Fig. 23 shows, operating the tubes with the bias at the cut-off point causes the plate currents to be half-wave pulses. Then (Fig. 24) the flux produced by the other will fill in the spaces. The combination produces the complete a.c. flux cycle f_c . In other words, one tube works on one half-cycle, and the other tube works on the other half-cycle in the class B pushpull stage.

The class B stage has an increased efficiency because each tube operates for only half of each cycle and passes practically no current the other halfcycle. Thus, there is no steady d.c. plate current to cause a constant power loss in the plate resistance of the tube. This is important because the amount of power that can be delivered from a stage depends on the ability of the tube to dissipate the heat developed within it. Thus, if a tube can dissipate 5 watts safely, we can increase the output only to the point where this power is lost in the plate resistance. Therefore, if we reduce the loss, we can safely increase the grid signal voltage to where the same loss occurs.

This is exactly what happens in a class B stage. Reducing the average plate power by causing the plate current to exist for only half the time permits the grid signal level to be increased (the grid can even go slightly positive). In turn, this produces high plate current pulses, so a large amount of a.c. power is delivered to the load. Since operation requires a specially designed voltage amplifier, the entire system must be designed for it. Fur-



FIG. 23. Class B operation is at the plate current cut-off point, so the plate current pulses exist for half cycles.

thermore, class B operation in the lowfrequency amplifier is possible ONLY in a push-pull circuit—otherwise half the signal would be destroyed.

► On paper, a class B stage looks much like a class A push-pull circuit. However, a fixed bias from the power supply is always used instead of selfbias, unless special zero-bias class B tubes are used. (Some class A stages use fixed bias also.) With a fixed bias instead of R_1 , Fig. 22 could be a class B stage, providing the following parts differences are considered:

Transformer T_1 will be a step-down transformer, so its secondary will have low reactance. This is necessary because the grid is driven positive part of the time in class B operation, causing a grid current flow. If the trans-



FIG. 24. As shown here, class B push-pull operation causes the tubes to supply current pulses alternately. The corresponding fluxes add to reproduce the entire a.c. cycle.

former had high reactance, the grid current flow through it would cause a voltage drop in the transformer, that would distort the peaks of the applied signal. Since the grids draw current, the grid circuit requires power—it is not just voltage-driven. Therefore, the preceding stage must be capable of furnishing this power, and the transformer T_1 must be an impedance-matching type. A low-power tube (in a class A circuit) is frequently used ahead of a class B amplifier. This tube is known as the "driver" tube, because it *drives* the class B stage to its normal operating point.

The class B amplifier delivers more power than is usually necessary in a home receiver. For this reason, true class B amplification is rarely found except in public address systems or in other applications requiring high power. However, class AB operation, in between class A and class B amplification, is found in many of the more powerful home receivers.

AB OPERATION

Class AB operation uses a C bias somewhere in between the cut-off bias of class B operation and normal class A bias. Then, if the driving voltage is limited so that the grid never becomes positive, the operation is known as class AB₁. On the other hand, if the signal voltage applied to the grid is permitted to go slightly positive (approaching class B operation) the operation is known as class AB₂. Class AB₂ operation delivers higher power than AB₁ but less power than does class B.

Inverse Feedback

The output of any amplifier using pentode or beam power tubes contains a considerable percentage of distortion. One effective method of reducing this distortion (and also reducing any excessive hum, oscillation, or noise) uses the principle of inverse feedback.

Feedback, as the name implies,

means feeding part of the output voltage of an amplifier back into its input circuit. If the feedback voltage arrives at the input exactly in phase with the original input voltage, we have positive feedback, often called *regeneration*. If positive feedback is carried far enough in an amplifier, the amplifier becomes

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unstable and finally oscillates.

For inverse feedback or degeneration, the phase relation between input and feedback voltages is such that the feedback voltage is made to arrive 180°_{out} of phase with the input signal. This means that whenever the input signal rises in a positive direction, the feedback voltage rises in a negative direction, and vice versa. The feedback voltage, therefore, always opposes the original input signal. The resultant input is the difference between the two voltages, so it is always smaller than the original signal.

Since inverse feedback *reduces* the effect of the original input voltage, it reduces the output also. This amounts to a reduction in the amplifier gain. (An important point to remember is that unlike positive feedback, which makes an amplifier unstable and sometimes causes oscillation, inverse feedback drops the gain and output, making the amplifier entirely stable and always under control.)

So far, the only result of inverse feedback we have seen is a reduction in gain. Let us see what benefits this sacrifice may offer.

REDUCTION OF DISTORTION

Let us examine the circuit in Fig. 25. Except for the inverse feedback connections, it is a conventional singleended power amplifier. Note that the lower end of input transformer T_1 is not returned to ground, but is connected to voltage divider R_1 - R_2 across the secondary of output transformer T_2 . (Resistors R_1 and R_2 are large in value compared to the impedance of the loudspeaker voice coil, so they absorb very little power from the circuit.) With this arrangement, part of the output voltage is fed back into the grid circuit; the exact amount of voltage can be regulated by adjusting the values of resistors R_1 and R_2 .

Now, what happens when an input signal is applied to the circuit? First, the input signal e_1 produces in the secondary of T_1 the grid voltage e_2 , which after amplification appears in the output as e_0 . For simplicity, let us assume that although e_2 is a pure sine-wave input voltage as in Fig. 26A, the output e_0 looks something like that in Fig. 26B. The "bump" on the output wave may represent noise, distortion, or hum (or all three) generated in the amplifier tube. The bump thus represents something that should be eliminated. because it was not present in the original signal.

Through our feedback circuit, a portion (e_3) of the distorted output e_0 is fed back to the grid in series with e_2 . This e_3 voltage is indicated by the dotted curves in Figs. 26B and 26C.

For our purpose, we can assume that all these things occur instantaneously;



FIG. 25. A typical single-ended power output stage utilizing inverse feedback. The portion of the output voltage across R₂ is fed back as e₃ into the grid circuit.

that is, there is no delay between the time e_2 is applied to the grid, and the instant that e_3 is fed back into the grid circuit.

Suppose we have connected the windings of output transformer T_2 so that feedback voltage e_3 arrives exactly 180° out of phase with input voltage e_2 , as shown in Fig. 26C. This

means we have inverse feedback. Since e_2 and e_3 are of opposite polarity, the resultant input is equal to their difference. This input is illustrated by the curve in Fig. 26D which is obtained by subtracting e_3 from e_2 in Fig. 26C.

The final output caused by the new reduced input will be something like that in Fig. 26E. It is apparent that



FIG. 26. These curves show how inverse feedback tends to correct amplitude distortion.

this is a much better reproduction of the original input than that obtained in Fig. 26B.

Through the use of inverse feedback we have taken the amplitude distortion components of the output and put them back into the input in such a direction that they tend to cancel themselves.

It is true that the output is reduced in amplitude in the process, because of the reduced effective input voltage (26D) is smaller than 26A, so 26E is less than 26B,) but if the input signal voltage $(e_1 \text{ in Fig. } 25)$ is increased, the output can be brought back to its former level while the distortion components remain substantially reduced.

The amount of reduction of all distortion, hum, and noise by inverse feedback depends upon the amount of output voltage fed back to the grid circuit and upon the gain of the amplifier without feedback. In general, the gain of an amplifier and the distortion in its output caused by the amplifier itself are reduced by an equal per cent. This means that if feedback voltage is increased until the gain has been reduced 50%, distortion will be reduced 50% also. If the input voltage is then doubled to return the output to its former value, the distortion in the stage will also double, but it will still be only 50% of what it would be without feedback.

Suppose we have an amplifier that without feedback, has the frequencygain characteristic curve 1 in Fig. 27. Let us say the poor response below 100 cycles is caused by coupling condensers that are too low in capacity, the peak at 3000 cycles by resonance effects in a transformer, and the loss of frequencies above 10,000 cycles by the shunting effect of tube capacities. Obviously, an amplifier with such a response would not give high-fidelity reproduction.

Now let us see what improvements inverse feedback might make. With a medium amount of output voltage fed back into the input in proper phase, we get response curve 2 in Fig. 27. This is a considerable improvement. Observe that although the over-all gain has been reduced, the low-frequency and high-frequency response is nearer the response of the intermediate range,





and the undesirable peak at 3000 cycles has been cut down.

How feedback brings this about may be visualized as follows: Below 100 cycles and above 10,000 cycles, where the amplifier's inherent gain is low, the output voltage also is low. This means the feedback voltage is small in magnitude, so the input is not reduced much. For the peak at 3000 cycles, however, the gain originally is high. The increased output voltage makes the feedback voltage quite large. When this is fed back, it cancels a large part of the original input and cuts the gain considerably. Therefore, the feedback corrects for frequency distortion as well as for amplitude distortion.

Even greater improvement in frequency response can be obtained if the value of the feedback voltage is increased, as in curve 3 of Fig. 27. Observe, however, that any increase in feedback is always accompanied by an additional reduction in gain. In-



FIG. 28. Inverse feedback from a special winding on the output transformer.

verse feedback may be increased only to the point where it is impossible to get enough driving voltage (or where it becomes uneconomical to go further).

► The remarkable ability of inverse feedback to reduce distortion has resulted in its widespread use in modern receivers that use pentode and beam power tubes, particularly those using single-ended stages. Let's study some of the typical circuits.

TYPICAL INVERSE FEEDBACK CIRCUITS

In addition to the basic circuit you have just studied, there are a number of variations on the method of getting the required feedback. It is generally



FIG. 29. Here the feedback is obtained from a voltage divider connected from plate to chassis.

unnecessary to feed back the entire output voltage to the grid circuit, and for this reason almost all feedback circuits use some form of voltage divider to reduce the feedback voltage. In the circuits shown in Figs. 28 to 32, the voltage divider resistors are marked R_1 and R_2 . The exact amount of feedback can be regulated by varying the resistance values.

► The output voltage developed across the average low-impedance loudspeaker voice coil (in a circuit like Fig. 25) does not always yield sufficient feedback voltage for satisfactory correction of distortion. However, an extra secondary winding having a greater number of turns can be added to the output transformer as in Fig. 28. This increases the feedback voltage with correspondingly better results.

► To get a high feedback voltage without an extra transformer winding, the voltage often is taken directly from the plate of the tube (Fig. 29). Condenser C_1 serves only as a blocking condenser to prevent short-circuiting the d.c. plate supply and prevent application of the plate voltage to the grid. The capacity of C_1 should be high enough so that the reactance of the condenser is small compared to the resistance of R_1 and R_2 , otherwise the feedback will not be the same for all frequencies.

The feedback voltage may be fed in *parallel* with the input instead of in series. This is usually done with resistance-coupled input circuits like that shown in Fig. 30. Note that R_2 is not only part of the feedback voltage divider circuit, but it is also the grid resistor for the tube. C_1 , as before, is only a blocking condenser.

► For feedback over two stages, the circuit in Fig. 31 is often used. Observe that the feedback voltage is inserted in the cathode circuit. The feedback could not be applied to the grid of the first tube because we have a phase



FIG. 30. The grid resistor R_2 divides the feedback voltage with R_1 .

shift of 180° in each tube, and with two tubes this makes a total shift between input and output of 360° . The voltage at the grid of the first tube, therefore, is exactly in phase with the output voltage, and coupling the feedback to the first grid would result in positive feedback. ▶ Feedback, however, can be applied to the grid of the first tube in a twotube amplifier if an extra output transformer winding is used. Fig. 32 is an example of this. In this case, the connections to the extra transformer winding may be reversed, if necessary, for proper phasing. R_3 serves only to load the transformer winding, and has no direct effect on the feedback voltage.

Current Degeneration. All the preceding systems are known as volt-



FIG. 31. Inverse feedback from the plate of the output tube to the cathode of a voltage amplifier.

age types, since it is the output voltage that is fed back. This system thus corrects the output voltage and hence reduces both frequency and amplitude distortion. There is another system which depends on the output tube plate *current*. This current type can correct for amplitude distortion but does not correct the frequency response.

A typical circuit is shown in Fig. 33. This is a conventional amplifier except that the bias resistor by-pass condenser has been omitted. With no condenser, the plate load consists of both the output transformer and the cathode resistor R_2 , and a small part of the output voltage is developed as e_3 across the cathode resistor. This voltage acts as a variable bias that opposes changes in the grid signal. When the grid voltage e_2 goes positive, the increase in plate current through R_2 increases the bias (e_3) and tends to make the grid more negative. When the signal swings negative, the bias e_3 is reduced. Hence, it has the same effect as the inverse feedback circuits already described.

Since this circuit acts to minimize changes in *plate current*, it can correct amplitude changes but not the frequency response, because the *same*



FIG. 32. Feedback from a special output transformer winding to the grid of a voltage amplifier.

plate current amplitude can cause differing output voltages if the load impedance changes with frequency.

FEEDBACK LIMITATIONS

It has been assumed that the feedback voltage is always 180° out of phase with the input voltage in these feedback circuits. Unfortunately, this is not always true. In addition to the 180° phase shift in each tube, there is always some phase shift in each coupling condenser or audio transformer. These extra phase shifts are quite small at medium audio frequencies, but at extremely high and extremely low frequencies they can become quite large. When more than two stages are included in the feedback path, it is possible that the net change in phase could result in positive instead of negative feedback, causing oscillation. Therefore, the design of these circuits must be carofully worked out. (It is necessary that any defective parts in these circuits be replaced by exact duplicates to avoid possible trouble of this kind.)

► All the inverse feedback systems shown here can be applied to push-pull stages as well. Incidentally, the cathode type of Fig. 33 introduces no degeneration in a push-pull stage (when the same resistor biases both tubes) unless the tubes become unbalanced. If the two tubes are the same, when one



FIG. 33. Current degeneration is obtained when the cathode by-pass condenser is omitted. This will correct for amplitude distortion but not necessarily for frequency distortion.

plate current is increasing the other is decreasing, so the drop across the bias resistor is practically constant. Therefore, there will be no degeneration until the circuit is upset and needs it the most! For this reason, you will frequently find that the bias resistor for push-pull tubes is not by-passed.

Phase Inverters

In our study of push-pull amplifiers, you undoubtedly noticed the use of two transformers. This was the standard circuit at one time. Modern practice, however, is to eliminate audio transformers whenever possible, since highquality ones are bulky and expensive, and cheaper ones do not give faithful frequency response.

We can't, of course, get rid of the impedance-matching *output* transformer. (Nor, in class AB_2 and class B operation, can we avoid using the input transformer, for no other device can perform its two functions of offering low impedance and of matching impedances to the driver.)

However, in class A and class AB_1 operation, the *input* push-pull transformer is an ordinary voltage amplifying transformer whose only special function is providing a way to feed the two grids 180° out of phase. It can be eliminated if we can find some other way of doing this.

As you know, a vacuum tube stage reverses the phase of a signal 180° because the signal voltage across the load is opposite in polarity to the signal applied to the grid of a tube. Therefore, if we feed the grid of one of the pushpull tubes from a voltage amplifier, and then put in another tube (called a *phase inverter*) to reverse the phase to feed the other push-pull tube, we can have push-pull operation (class A) without using an input transformer.

A resistance-coupled stage is used as the phase inverter. Its use offers several advantages: higher fidelity characteristics, a reduction of the possibility of magnetic coupling and hum pickup, and a reduction in weight of the equipment. In general, it is cheaper to use a tube stage than the relatively expensive input push-pull transformer. Of course, in addition to reversing the phase, the additional stage must not feed a greater signal to its pushpull tube than is fed to the other. Therefore, it is necessary to limit the gain so that the inverter tube will not have greater output than the voltage amplifier.

▶ Fig. 34 shows a simplified diagram of an inverter. VT_1 is a voltage amplifying stage. It amplifies the signal, and feeds tube VT_2 through the resistance coupling R_3 - C_4 - R_4 - R_5 . A voltage di-



FIG. 34. A basic phase-inverter stage.

vider R_4 - R_5 is used to feed a portion of the output voltage from VT_1 to the phase inverter tube VT_4 .

The tube VT_4 amplifies in the normal manner. However, the voltage divider R_4 - R_5 is figured out so that if this stage has a gain of, say, 10, then only one-tenth of the voltage across the R_4 - R_5 combination is fed to the tube. Hence, there is effectively no gain in the VT_4 stage (10 times 1/10 is 1), and the voltage across R_9 is the same amount as the voltage across R_4 - R_5 , so tube VT_3 gets the same amount of signal voltage as does tube VT_2 . (When servicing this type of phase inverter, it is important to replace R_4 and R_5 with

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exact duplicate parts when these resistors are found defective. This is necessary so that the proper voltage division will be obtained to counteract the gain of VT_4 .)

What about the phase? When the signal in the grid circuit of VT_2 is going positive, this same signal is fed to the grid of VT_4 . The phase reversal of VT_4 causes the grid of VT_3 to be driven



FIG. 35. The phase inverter and the driver are sections of the same dual tube.

negative at this same moment—just what we want to happen.

▶ Fig. 35 shows another arrangement using the circuit of Fig. 34. Tubes VT_1 and VT_4 are individual sections of a dual-triode tube. The same bias resistor (R_1) is used for both sections, a practice that tends to equalize any slight irregularities in the characteristics of VT_1 and VT_4 .

DEGENERATIVE PHASE INVERTERS

There is danger that aging of the tubes may upset the voltage balance between the outputs of VT_1 and VT_4 even when they are part of the same tube. If an unbalanced voltage is fed into the push-pull circuit, we will get a distorted output, as the even harmonics will not balance out.

One way around this is to use a single tube to feed both the push-pull tubes, as shown in Fig. 36.

First, let's examine the circuit of VT_1 . Tracing the direction of electron

flow from cathode to plate, we find that the end 3 of R_4 is negative with respect to terminal 4. Continuing around through the B supply, and through R_3 back toward the cathode, we find that terminal 2 of R_3 is negative with respect to terminal 1.

Now, let's consider terminal 4 of R_4 to be grounded through C_4 (insofar as a.c. signals are concerned), so that it connects effectively to terminal 2 of R_3 . Comparing the voltages across these two resistors, we find that an increase in plate current makes terminal 3 of R_4 more negative at the same instant that it makes terminal 1 of R_3 more positive. Therefore, if we feed VT_2 from terminal 3 of R_4 , and feed tube VT_3 from terminal 1 of R_3 , the two tubes will be fed out of phase. R_3 and R_4 must be equal in size, of course, so that the voltages fed the push-pull tubes will be equal.

Since R_3 is part of the plate load, but is also in the input circuit in such



FIG. 36. Both push-pull tubes may be fed from a single tube in this unique phase inverter stage.

a way that the voltage developed across R_3 opposes the input signal voltage, this circuit is degenerative. For example, if the stage gain is 50, then 1 volt across R_1 (the grid resistor) will appear as 25 volts across R_4 and R_3 respectively. (25 plus 25 is 50, the a.c. voltage we would get from a gain of 50.) However, the 25 volts across R_3 must be subtracted from the voltage eapplied at the input terminals, so to get 1 volt across R_1 , we must apply an e of 26 volts. Hence, the over-all gain of VT_1 is only about 2 (50 \div 26 is about 2). As only half the output voltage is supplied to each of the push-pull tubes, the effective gain is one, so the input voltage e we supply must be the same as the voltage necessary to drive the grid of one of the push-pull tubes.

As the tube VT_1 ages, the voltage across R_3 and R_4 will tend to decrease, but this lets a slightly increased amount of voltage appear across grid resistor R_1 , and conditions are restored approximately to normal. Furthermore, even if the input voltage increases to large amounts, the resulting increase in voltage across R_3 prevents overload. Thus, all the advantages of degeneration are added to this phase inverter stage.

► The same idea of degeneration is also used in what is called a self-bal-



FIG. 37. By making a slight change in the grid feed circuit of the phase inverter stage, a balanced phase inverter, including degeneration, is obtained.

ancing inverter using two tubes. The circuit is shown in Fig. 37. At first glance, it appears to be the same as that of Fig. 35, but notice the position of resistor R_6 . This resistor is placed so that degeneration occurs, and its value is increased to equal that of R_5 (and R_7).

Half the voltage between point 1 and ground (point 1 is the grid terminal of VT_2) is applied to the grid of VT_4 . The amplified plate voltage of VT_4 then appears between point 2 and ground; half of it is across R_6 , and is fed back into the grid circuit of VT_4 . Since (as in the circuit of Fig. 36) the output voltage is fed back to the grid of VT_4 , the effective gain is only 2 in-



FIG. 38. A special phase inverter circuit used on certain Philco receivers. One of the output tubes is used as the source of the grid voltage for the other.

stead of being the natural gain of the stage (and, since VT_4 gets only half the R_5 - R_6 voltage, the gain from point 1 to point 2 is only one—the two voltages are equal). However, this gain is practically independent of the characteristics of tube VT_4 , so the tube can age without upsetting the circuit.

► A somewhat unique phase inverter circuit is shown in Fig. 38. This is used in certain Philco receivers.

Here, the input signal e is fed directly to the grid of one of the pushpull tubes (VT_1) through condenser C_1 . The grid resistor for this tube is R_1 . This tube amplifies the signal in the normal way.

Inserted in the screen grid circuit of tube VT_1 is a small resistor R_4 . This acts as a load resistor, in that a certain amount of the signal energy developed by VT_1 appears across this resistor. In other words, the screen grid acts like a triode plate, and taps off a certain amount of the signal energy, with resistor R_4 acting as its load. The signal developed across R_4 is fed to the grid of VT_2 through C_2 . Resistor R_3 is the grid resistor for this tube.

Thus, the tube VT_1 serves a dual purpose. It is one of the tubes in the push-pull amplifier, and-also acts as the phase inverter for tube VT_2 .

Considerable degeneration is necessary in this circuit to prevent distortion. This is furnished by resistor R_2 , which is a resistance of higher value than either R_1 or R_3 . Through it, some of the energy fed back through C_2 is applied to the grid of VT_1 , opposing the original signal and causing degeneration in this tube. In addition, some of the input signal e is fed through R_2 to VT_2 so that it is out of phase with the C_2 voltage. Hence both tubes undergo a certain amount of degeneration. However, degeneration is not carried as far as in the phase inverters we previously described, because nearly normal performance is wanted from VT_1 . Just enough is used to give reasonable fidelity.

► There are a number of other less commonly used phase inverter circuits. However, they all work essentially like the basic ones we have just described, so you will have little trouble in understanding them.

Facts About Low-Frequency Amplifiers

This lesson is an important milestone in your Course. Here, you gain a complete understanding of one of the major kinds of amplifiers. This lesson might also be considered a summary of vour previous studies-because vou use facts from all the earlier lessons. You put together your knowledge of how condensers and tubes work, facts you have learned about basic amplifiers, your knowledge of sound waves and their characteristics, and your knowledge of the kinds of distortion. So many important facts are in this lesson that you should come back to it from time to time and quickly run through it to refresh your mind on important circuit and stage actions.

► As a summary of the highlights of this lesson, the following facts will be helpful. If any are not clear in your mind, then it is advisable for you to go back over the lesson right now, because you will need these important facts for later lessons.

The low-frequency amplifier deals with the intelligence signal—audio or video. In sound receivers, the audio signal must be increased in power to operate the reproducer properly, so a power output stage is used. However, power stages require considerable voltage to drive them, so voltage amplifiers are used to increase the output of the demodulator so that it can operate the power output stage properly.

The voltage amplifiers are named according to the devices used to couple the stages together (or to couple to the power output stage). Hence, voltage amplifiers are known as resistancecapacitance-coupled, impedance-coupled, and transformer-coupled stages.

Of these, the resistance-capacitancecoupled stage offers low gain but is capable of the best fidelity characteristics. It is possible to get a frequency range wide enough to make frequency distortion of minor importance. Amplitude distortion can be kept within reasonable limits also, and phase distortion is of importance only in television. (You will learn more about trouble with phase distortion as you study television receivers in detail.)

When the voltage amplifier has built up the voltage sufficiently, this voltage is used to drive a single-ended or a push-pull power output stage.

 \blacktriangleright In the single-ended power stage, a single power tube (or tubes in parallel) will be found. The triode offers the best fidelity, but it requires a much higher signal voltage. For this reason, most modern receivers use pentode or beam power tubes. Both the pentode and the beam power tubes have higher power sensitivity than the triode (they require a much smaller grid signal voltage). For example, earlier radio receivers used two or sometimes three voltage-amplifier stages to drive a single triode power output stage. With modern receiver design, a somewhat greater output is obtainable from the demodulator, so it is possible for a single voltage-amplifier stage to give enough voltage to drive pentode or beam power stages.

You may wonder why both pentode and beam power tubes are used when they seem so similar. The beam power tube is somewhat more efficient than the pentode, and does not introduce quite as much third-harmonic distortion because its characteristic curve is shaped in a slightly different way. The pentode tube was introduced first, and is still used for low-power outputs. However, the more efficient beam power tube has replaced the pentode in the majority of receivers.

In power output stages, the high frequencies may be cut down somewhat by the necessity for a by-pass condenser across the plate load, and the low-frequency response depends upon the primary inductance of the output transformer. Even so, it is possible to make the frequency response of these stages fairly good.

However, the pentode and the beam

power tubes introduce considerable amplitude distortion. In the singleended stage, much of the amplitude distortion can be corrected by inverse feedback, and the push-pull output stage has the important advantage of eliminating all even-harmonic distortion that originates in the output stage. Inverse feedback can be used for push-pull stages also, and will cut down the odd-harmonic distortion.

► In order for a push-pull stage to function properly, the grids of the two tubes must be fed 180° out of phase. This requires a special input transformer or a phase inverter tube stage. The input transformer has been practically eliminated from modern receivers because of its undesirable characteristics and rather high cost. Instead, a phase inverter is almost universally used to feed the grid of one of the push-pull tubes.

The two push-pull tubes must be fed with input voltages of the same amplitude. When a phase inverter stage is used, one of the push-pull tubes is fed directly from a voltage amplifier stage. A fraction of the output of this stage is also fed to the phase inverter stage. There this fractional voltage is amplified and inverted in phase; then it is fed to the other push-pull tube. The amplification received in the phase inverter stage is just enough to make the two push-pull grid voltages equal. (A single tube may feed both push-pull tubes if its load is split between its plate and cathode circuits so that phase inversion occurs.)

► Inverse feedback is a system in which part of the output energy is fed back to some point in an amplifier where it will be out of phase. Any distortion or other unwanted component that originates between these points (the output and the point where the feedback is returned) will be fed back

out of phase and thus will tend to cancel itself. Therefore, inverse feedback is particularly used with pentode and beam power tubes to correct their amplitude distortion. Of course, inverse feedback lowers the gain of the stage or stages within the feedback path, so a higher input voltage must be supplied at the input terminals. For this reason, inverse feedback is usually limited to the power output stage and perhaps to one preceding stage. Another voltage amplifying stage must be used ahead of the inverse-feedback circuit to deliver the required input voltage. Naturally, amplitude distortion must be controlled in this voltage amplifier.

The results of inverse feedback are such that a beam power output stage utilizing this circuit can deliver practically the same quality of output as the triode tube and still require somewhat less input voltage than would a triode tube giving the same power output.

▶ In modern radio receivers, you will recall that there are many kinds of power supplies. Some receivers operate from the power line and use a power transformer. It is almost universal to find a push-pull stage in such sets when they are intended to give good fidelity. On the other hand, if the receiver is an a.c.-d.c. type or is battery operated, there is a limit to the amount of voltage and current available. For this reason, such sets commonly use singleended power output stages and smaller loudspeakers, which require less power.

In a television set, there may be from one to three voltage-amplifier stages. The television cathode ray tube —the tube on whose face the picture is reproduced—is the "output" stage in a television set. This tube is entirely voltage driven—very little power is needed to moduate the electron stream within it. Therefore, in a television set, the low frequency (or video) amplifier consists primarily of voltage amplifying stages.

► As a radio serviceman, you are interested in all these stages because the parts are subject to natural breakdowns—condensers open, short-circuit, or change in value; resistors open or burn out; transformers short-circuit and open-circuit; and tubes change their characteristics with age. You need to know just how these amplifiers operate so that you can intelligently choose the proper replacement part.

Of course, when you have the manufacturer's instructions, or the part values are marked, you need only use parts like the originals to get the same results. However, there will be plenty of instances in which you will have to make a replacement and will not know the exact values of parts. Then you can be guided by your knowledge of what will happen if you use a resistor or a condenser that is larger or smaller than the original. Also, your knowledge of how standard circuits work will always be helpful when you meet non-standard circuits.

You are now in a much better position to judge the fidelity that would be expected from different resistance-coupled amplifiers. However, always remember that some manufacturers make up for the deficiencies of one stage by purposely changing the response of another. Therefore, don't make the mistake of assuming that the response of the amplifier is necessarily poor because the design of one stage happens to sacrifice low frequencies or high frequencies. Another stage may have an excess response to these frequencies, so that the final result of the combination may be a relatively good output.

Lesson Questions

Be sure to number your Answer Sheet 15FR-2.

Place your Student Number on every Answer Sheet.

Send in your set of answers for this lesson immediately after you finish them, as instructed in the Study Schedule. This will give you the greatest possible benefit from our speedy personal grading service.

- 1. Does the low-frequency amplifier obtain its input signal directly from: 1, the demodulator; 2, the i.f. amplifier; or 3, the power supply?
- 2. If the low-frequency response of an R-C coupled amplifier is to be increased, should the capacity of the coupling condenser $(C_3 \text{ in Fig. } 1)$ be: 1, increased; or 2, decreased?
- 3. If a stage having a gain of 12 is cascaded with a stage having a gain of 10, what will be the total voltage amplification of the two?
- 4. Will an increase in the ohmic value of the load resistance for a triode tube make the dynamic E_{g} -I_P curve for the tube straighter (more linear)?
- 5. With a triode tube, is the maximum undistorted power obtained when the plate load: 1, is equal to; 2, is one half of; or 3, is twice the plate resistance?
- 6. What is the purpose of the plate-to-cathode condenser (C_3 in Fig. 14) that is used in power output stages?
- 7. Does the push-pull stage eliminate: 1, even; or 2, odd harmonic distortion?
- 8. If there is no by-pass condenser across the bias resistor in the cathode circuit, is the inverse feedback: 1, the voltage type; or 2, the current type?
- 9. What is the purpose of the phase inverter stage?
- 10. Why is it important to replace resistors R_4 and R_5 in Fig. 34 with exact duplicate parts?

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HOW TO START STUDYING

For some people, starting to study is just as hard as getting up in the morning. An alarm clock will work in both cases, so try setting the alarm for a definite study-starting time each day. Start studying promptly and definitely, without sharpening pencils, trimming fingernails or wasting time in other ways.

Beginning is for many people the hardest part of any job they tackle. So formidable does each task appear before starting, that they waste the day in dilly-dallying, in day-dreaming, and in wishing they didn't have to do it. The next day, and the next after that are the same story. Indecision brings its own delays, making it harder and harder to buckle down to work.

Are you in earnest? Then seize this very minute; begin what you can do or dream you can. Boldness in starting a new lesson is a great moral aid to mastery of that lesson; only begin, and your mind grows alert, eager to keep on working. Begin, and surprisingly soon you will be finished.

J. E. SMITH





STUDY SCHEDULE NO. 16

For each study step, read the assigned pages first at your usual speed, then reread slowly one or more times. Finish with one quick reading to fix the important facts firmly in your mind, then answer the Lesson Questions specified for that step. Study each other step in this same way.

T.R.F. Receivers; Superheterodyne Receivers; Side Bands; Fidelity; R.F. Amplifiers in Transmitters; What Is a Resonance Curve; Typical R.F. Amplifier Stage; How a Resonance Curve Is Secured; Defining Selectivity; Amplifiers in Cascade; Ideal and Practical Ratio Resonance Curves. Answer Lesson Questions 1, 2, 3 and 4.

Amplifier Loads; Coil-Condenser Loads; Coil-Coudenser Load Connections; Tuned Transformer Circuit; Inter-Electrode Capacities; Practical Considerations. Answer Lesson Questions 5 and 6.

3. Getting Rid of Feed-Back; The Screen Grid Tube; The Suppressor Grid TubePages 15-20

Grid Suppressors; Neutrodyne Circuits; The Performance of a Screen Grid Tube; purpose of the suppressor grid; suppressor grid circuits; characteristic curves for tubes using suppressor grids. Answer Lesson Question 7, 8 and 9.

Cross-Modulation; Modulation Distortion; grid construction in a variable-mu tube; mutual conductance and plate current curves; r.f. amplifier circuits using peutode tubes. Answer Lesson Question 10.

Problems of flat response; double-peak response; band width; purposes of fixed r.f. amplifiers; power r.f. amplifiers for transmitters; Efficiency; class A amplifier; class B amplifier; class C amplifier; Self-Biased R.F. Amplifier Circuits; Power Amplificatiou; Pusb-Push Power R.F. Amplifiers.

- 6. Mail your Answers for this Lesson to N.R.I. for Grading.
- 7. Start Studying the Next Lesson.

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Radio Frequency Amplifiers for Sound and Television Communication

WHAT R. F. AMPLIFIERS DO

I sound or television receivers the modulated R.F. carrier current in the antenna circuit is usually not strong enough to be fed directly to the detector (where the sound or picture signals are separated from the carrier). Weak modulated carrier voltages must often be amplified as much as a million times before they can be satisfactorily demodulated by the detector; this can be done by an R.F. amplifier consisting of one or more stages.

This R.F. amplifier must not only *amplify* the desired incoming carrier signal, but must also be able to reject all undesired carrier signals which are present in the antenna circuit; the R.F. amplifier must have selectivity. Furthermore, the R.F. amplifier must be able to select, when the receiver controls are adjusted, any one of the many carriers which may be present in the antenna circuit; it must be possible to *tune* the R.F. amplifier to a desired carrier frequency.

T.R.F. Receivers. In one very common type of R.F. amplifier, all of the R.F. stages can be tuned to the frequency of the desired incoming carrier signal. A receiver using this type of R.F. amplifier is commonly known as a tuned radio frequency receiver, or simply as a T.R.F. receiver.

Superheterodyne Receivers. The most widely used R.F. amplifier is that in which only the first stage (or first two stages) is tuned to the frequency of the incoming carrier. The amplified carrier signal from the first stage then enters a mixer-first detector tube, where it is mixed with an R.F. signal from a local oscillator in such a way that the frequency value of the R.F. carrier is reduced to a lower R.F. value which is known as the intermediate frequency (I.F.). This intermediate radio frequency, still modulated with either the sound or picture signal, is then amplified by I.F. amplifier stages having fixed tuning and therefore giving maximum amplification at one particular frequency. This is the so-called superheterodyne (or super) principle of R.F. amplification, in which the tuning action takes place before frequency conversion and most of the amplification takes place after frequency conversion (in the I.F. amplifier stages, which are also known as low R.F. amplifier stages). Each R.F. and I.F. stage contributes something to the selectivity of the receiver, with the greatest amount of selectivity being secured in the fixed tuned stages (the I.F. amplifier).

Side Bands. Let us review briefly the characteristics of a modulated radio frequency carrier current. As you know, the process of modulating a carrier introduces side frequencies which are above and below the carrier frequency value. The greater the frequency range of a sound or picture signal being transmitted, the farther off from the carrier frequency will be the extreme side frequencies. The frequency range extending from the lowest to the highest side-band frequency is known as the band width; this naturally varies with the nature of the intelligence being transmitted. In addition to the carrier frequency, then, the R.F. amplifier must therefore handle all of these side frequencies.

Fidelity. The R.F. amplifier in a receiver must be able to reject all signals except the desired R.F. carrier and its side bands, but at the same time it must not be so selective in its operation that it completely or partially rejects any of the side frequencies associated with the desired R.F. carrier frequency.

The fidelity (quality of reproduction) of an R.F. amplifier is impaired when the carrier frequency is amplified more than the side frequencies; distortion likewise occurs if the side frequencies higher than the carrier are amplified more than the side frequencies which are lower than the carrier.

The R.F. amplifier in a sound or television radio receiver must therefore be able to do these four things: 1, tune to the desired carrier frequency; 2, amplify the desired carrier frequency; 3, contribute selectivity by rejecting undesired carrier frequencies; 4, give uniform amplification for all side frequencies when high fidelity reproduction is required.



FIG. 1. Simplified schematic circuit diagram of a typical R. F. amplifier stage.

R.F. Amplifiers in Transmitters. In any radio transmitter, regardless of whether sound or picture signals are handled, the R.F. amplifier can be divided into two sections: 1, the section handling only the unmodulated R.F. carrier currents; 2, the section handling the modulated R.F. carrier current. The point of division is therefore the stage in which modulation takes place.

The first section of a transmitter creates, by means of an oscillator stage, an R.F. carrier which is amplified by succeeding R.F. stages until the R.F. carrier current is at the power level required for modulation purposes. Since only a single frequency passes through this first section, the R.F. stages may be very selective.

The process of modulation introduces side frequencies as well as harmonics. The second section of a transmitter must therefore pass a definite band width, amplifying all frequencies in this band equally well while raising the power of the modulated R.F. carrier to the desired value.

RESONANCE CURVES

What Is a Resonance Curve? The manner in which the amplifying ability of a tuned R.F. amplifier stage varies for the different frequencies between the lowest and the highest side frequencies is of great importance; with this information at hand, we can tell how much the stage amplifies. how well it rejects undesired signals, how it amplifies desired signals, and how it amplifies some frequencies more than others in the side bands associated with the desired carrier. All this information about a tuned R.F. amplifier can be presented with a frequency response curve which shows in graphical form the manner in which the amplifier handles or amplifies the various frequencies in its operating range. Since tuned R.F. amplifiers alwavs contain at least one resonant circuit, this frequency response curve is often called a resonance curve.

Typical R.F. Amplifier Stage. The diagram in Fig. 1 presents in simplified form a typical tuned R.F. amplifier stage; after reviewing briefly the



Courtesu General Radio Co.

FIG. 2A (left). General Radio type 605-A signal generator, which consists (undamentally of an R.F. oscillator having a frequency range of 95 kc. to 30,000 kc., and means for varying and measuring the output voltage. There are also provisions for modulating the R.F. signal with an audio note of faced frequency or with an external modulating signal when a modulated carrier is required; the meter on the panel indicates percentage modulation in this case.
FIG. 2B (right). General Radio type 726-A vacuum tube voltsmeter, a multiple-range instrument designed to measure R.F. voltages from .1 to 150 volts with a high degree of accuracy. It is calibrated to read r.m.s. (effective) values of sine wave voltage.

operation of this stage, I will show you how a resonance curve for it can be obtained.

The R.F. source in Fig. 1 feeds to the grid-cathode of the tube (between points 1 and 2) an R.F. voltage which we will call e_g . As a result of the amplifying action of the tube and the effects of the load, there appears across the output terminals of the load an R.F. output voltage which we will designate as v_{o} . The net or true voltage amplification of this R.F. amplifier stage (expressed as A) is the output voltage v_0 divided by the input voltage e_{g} . The type of tube used, the operating voltages and the type of load applied determine to a great extent the behavior of this stage as an amplifier.

How a Resonance Curve Is Secured. Although an R.F. amplifier must handle many frequencies (the carrier frequency and the side-band frequencies) simultaneously in actual practice, it is obvious that we cannot study the performance of the amplifier while all these frequencies are pass-

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ing through; the practical way is to send through only one frequency at a time, a simple sine wave R.F. signal, and measure how much it is amplified. The frequency of this test signal is then changed to other values, and the amplification is measured at each frequency.

Radio engineers use a signal generator like that shown in Fig. 2A to supply a sine wave R.F. Signal whose frequency can be adjusted to the various values required when testing an R.F. amplifier. The large dial in the upper right corner of the panel controls the frequency of the output, the dial at its left changes the frequency range, and other dials change the strength and the modulation percentage of the R.F. output signal when a modulated signal is desired.

To measure the input and output voltages of an R.F. amplifier, engineers use a special voltmeter which employs a vacuum tube as a detector; a typical laboratory type vacuum tube voltmeter (abbreviated V.T.V.M.) is shown in Fig. 2B. The voltage being measured is applied to the wedge-shaped probe (at the left of the cabinet) in which is an acorn-type tube connected as a rectifier. A shielded cable connects the probe unit to the cabinet on which is mounted a multi-range voltmeter and the control switches. The amplification produced by an R.F. stage at any frequency can be determined simply by dividing the measured output voltage v_o by the measured input voltage e_a .

If the R.F. amplifier stage in Fig. 1 is set to amplify a 1,000 kc. signal, and we desire to obtain a resonance curve for the amplifier at this resonant frequency, we would connect the signal generator in place of the R.F. source and set it in turn at 980, 985, 990, 995, 1,000, 1,005, 1010, 1,015 and 1.020 kc., and for each setting connect the vacuum tube voltmeter across terminals 1 and 2 to measure e_z (if its value is not already known) and across terminals 3 and 4 to measure v_0 . We would then compute the amplification A obtained at each frequency, and plot our information on graph paper in the manner shown in Fig. 3 to secure the resonance curve for the R.F. amplifier at 1,000 kc. Notice that as the frequency is increased from the lowest value, the amplification increases and reaches a maximum value at 1,000 kc., the resonant frequency of the stage. Further increases in frequency beyond 1,000 kc. give decreasing amplification, the amplification dropping to zero at about 20 kc. above the resonant frequency. A curve of this nature is called a single-peaked resonance curve, for it gives maximum or peak amplification at a single R.F. value.

When a 1,000 kc. R.F. carrier is modulated with a sound signal having a maximum frequency of 5 kc., you know that the side-band frequencies will extend from 995 to 1,005 kc. The amplifier represented by the response curve in Fig. 3 would amplify the carrier frequency about 65 times, but would amplify the lowest side frequency $(f_1 \text{ in Fig. } 3)$ only 50 times and the highest side-band frequency (f_2) 58 times.

An amplifier which has a single peaked resonance curve distorts a modulated carrier in two ways. If the corresponding upper and lower side frequencies are amplified equally well but less than the carrier frequency, the result after demodulation is comparable to frequency distortion (where the extremely high frequencies in the modulation signal are eliminated). If the corresponding upper and lower side frequencies are, in addition, amplified unequally, as is the case in Fig. 3, there is amplitude distortion as well.

Defining Selectivity. Since the U.S.A. radio stations in the broadcast band are only 10 kc. apart in frequency, it is essential that a receiver tuned to one station does not also pick up the stations 10 kc. away on either side; insufficient selectivity in an R.F. amplifier therefore results in station interference. If the tuned R.F. stage whose performance is represented by Fig. 3 is the only one in a receiver, and it is tuned to a 1,000 kc. station, its carrier signal would be amplified about 65 times, while the carrier signals of adjacent stations (at 990 kc. and 1,010 kc.) would be amplified 15 times and 28 times respectively; this is clearly poor selectivity, for all three stations, if received with nearly equal signal intensity, will be heard at once. What, then, constitutes good selectivity?

Selectivity was formerly found according to the following procedure: The two frequencies at which amplification was 70% of the maximum amplification (the value at resonance) were found; the difference between these frequencies was determined, and the resonant frequency divided by this difference in frequency was said to be a measure of the selectivity of



the amplifier. Points a and b in Fig. 3 correspond to the two frequencies at which A is 70% of its maximum value. The difference in frequency between these two points is 13.5 kc., hence the selectivity value for this particular amplifier stage would be 1,000 (the resonant frequency) divided by 13.5, or 74. Incidentally, this value is also the Q factor for the resonant circuit of this amplifier stage.* This older definition for selectivity and Q factor is perfectly satisfactory when applied to simple or single resonant circuits, since all such circuits have essentially identical resonance curve shapes. The response curves of complicated or multi-resonant circuits have widely varying shapes, however, making this old definition of selectivity unsuited for present day use.

The modern definition for receiver selectivity is based upon the fact

^{*} In a series resonant circuit which is tuned to resonance, the ratio of the voltage across the coil or condenser to the applied circuit voltage is the Q factor of the circuit; Q factor is also the reactance of the coil divided by the circuit resistance. In a parallel resonant circuit the ratio of the coil or condenser current to the line supply current at resonance is the Q factor; Q factor is also the resonant resistance divided by the coil or condenser reactance.
that we are primarily interested in how much better the amplification is at resonance than at the nearest undesired frequencies. For good selectivity, the desired signal frequency and its side frequencies must be amplified at least 1,000 times more than the nearest undesired signal and its side bands. For fair selectivity this ratio (which we shall designate as R_A) can be about 100, and for excellent selectivity the ratio must be higher than 10,000.

Fidelity is now defined in much the same way as selectivity. For good fidelity the desired carrier frequency must not be amplified more than 1.25 times the highest side frequency desired with that signal; in other words, amplification at the highest side frequency must be at least 80% of the amplification at resonance. These figures are by no means fixed, for they are based upon opinions rather than upon precise data, and in addition will vary with the purpose of and use to which an R.F. amplifier is put.

A better picture of the frequency response characteristic of an R.F. amplifier can be obtained by computing R_A , the ratio of amplification at resonance to amplification at the test frequency, for each value of test frequency and plotting these ratios against frequency in the manner shown in Fig. 4.* A curve like this (called a *ratio resonance curve*) gives us both the fidelity and selectivity characteristics of an amplifier; for example, in Fig. 4 the highest side frequency (for a 5 kc. modulating signal) has a ratio of about 1.1 (at 1,005 kc.), which indicates excellent fidelity since it is considerably less than the maximum allowable value of 1.25. At 1,010 kc., however, the ratio is only about 2.3, indicating that the desired carrier frequency signal will be amplified only about 2.3 times more than the carrier signal of an undesired station at 1,010 kc. if both carrier signals have equal signal strength at the location of the receiver. Thus, fidelity is good but selectivity is very poor in the amplifier represented here.

Amplifiers in Cascade. If one R.F. amplifier stage raises the voltage at the resonant frequency 10 times, another similar stage connected in cascade will amplify the voltage 10 times further, giving a total amplification of 100; the addition of a third identical amplifier stage will boost the voltage 10 times further, giving a total amplification of 1,000. The amplification (gain) of an R.F. amplifier is therefore the product of the gain of each individual stage (the gains are multipled together).

Resonance curves for two and three tuned R.F. amplifier stages in cascade are given in Fig. 5, with the curve for a single stage shown for comparison. The addition of stages makes the resonance curve sharper and higher in peak value. Cascading of R.F. amplifier stages thus boosts the amplification and improves the selectivity of an R.F. amplifier,

The resonance curve for perfect fidelity and ideal selectivity should be flat throughout the desired band width and should have straight vertical sides; adding stages makes the sides of the resonance curve more nearly vertical, giving better selectivity, but lowering fidelity (by making the resonance curve sharper in the band width).

^{*} Laboratory engineers sometimes measure R_{\perp} directly. At each test frequency they increase the output of the signal generator until the measured amplifier output voltage v_{\circ} is the same value as it was at the resonant frequency; the signal generator output at the test frequency divided by the signal generator output at resonance gives the value of R_{\perp} .

Fidelity and selectivity characteristics of amplifiers in cascade are more clearly presented by plotting R_{\star} against frequency, as was done in Fig. 4; this is done in Fig. 6. Let us analyze these three ratio resonance curves; for convenience I will set down again the practical definitions for selectivity and fidelity.

Good fidelity: R_{A} is lower than 1.25 at the highest side frequency.

Good selectivity: R_{A} is higher than 1,000 on both adjacent-channel carrier frequencies.

Checking fidelity first, we find that good fidelity is secured with one stage ($R_{\rm A}$ is less than 1.15 at 1,005 kc.) and with two stages ($R_{\rm A}$ =1.25),







Fig. 6. These three ratio resonance curves give the fidelity and selectivity characteristics of one, two and three of the R.F. amplifier stages in Fig. 1. The scale at the left gives values of R_A (amplification at resonance + amplification at test frequency). Resonance is at 1,000 kc.

but three stages give poor fidelity ($R_{\rm A}$ is higher than 1.25). As to selectivity, $R_{\rm A}$ for three stages is about 100 at 990 kc. and is about 10 at 1,010 kc., these being the adjacent-channel carrier frequencies in the broadcast band. The lowest value governs, and hence selectivity is poor. For one and two stages, the values of $R_{\rm A}$ at these undesired carrier frequencies are lower, so selectivity is even poorer than for three stages.*

^{*} When peaked R.F. amplified stages have symmetrical response curves, several stages may be connected in cascade to give good selectivity, and the attenuation of the higher modulation frequencies (the poor frequency response) can be compensated for by using an audio (or video) amplifier which will give greatest amplification at the higher frequencies. This fact makes it possible to equalize (adjust) a receiver for high fidelity even though individual stages in it do not have ideal characteristics.

You will notice that the vertical scales in Figs. 5 and 6 are arranged in a different manner from those in Figs. 3 and 4; this "condensing" was done in order to cover a wide range of values while still showing the nature of the curves clearly at the lowest values, around the resonant frequency. These condensed scales are known as logarithmic scales; they also tell us more clearly what the ear is going to hear, for our ear has this same logarithmic response.

Complete R.F. amplifiers, regardless of whether they are of the superheterodyne or the T.R.F. type, are measured in exactly the same manner as the single stage shown in Fig. 1, and the results are plotted exactly as is done in Figs. 5 and 6. The resulting curves allow the radio engineer to study the amplification, selectivity and fidelity characteristics of the entire R.F. amplifier.

Ideal and Practical Ratio Resonance Curves. Ratio resonance curves for four different R.F. amplifiers, each of which contains several stages in cascade, are given in Figs. 7A and 7B. These curves give the value of the ratio $R_{\rm A}$ for different frequencies up to 20 kc. above and below resonance. You can consider $R_{\rm A}$ in two ways, as the ratio of the amplification at resonance to the amplification at various off-resonance frequencies, or as how much stronger the signal must be off resonance to produce the same output as at resonance. The higher the value of $R_{\rm A}$ for frequencies off resonance, the more selective is the amplifier.

Before you can study the ratio resonance curve for a particular R.F. amplifier, you must know at least two facts: 1, the frequency separation in kc. between the stations which are to be picked up by that amplifier; 2, the R.F. band width in kc. which is required to handle the entire range of frequencies for the audio or video signal being transmitted.

Stations in the broadcast band in the United States, radiating sound signals, are located 10 kc. apart and always have a frequency value ending in zero; thus you may receive stations at 570, 580, 590 kc., etc. For sound signals having a maximum range of 5,000 cycles, the side frequencies will extend 5 kc. above and below the carrier frequency. For good selectivity, then, the value of R_A should be greater than 1,000 at frequencies more than 5 kc. off the desired carrier frequency. To secure good amplification of the side frequencies, however, R'_A should be as close to the value 1 as possible throughout the band width of 10 kc.

Curve 1 in Fig. 7A (the dash-dash curve) represents the ideal ratio resonance characteristics for an R.F. amplifier designed for use with broadcast band apparatus; all frequencies in the band width are amplified equally, while all other frequencies are completely cut off. Curves 2 and 3 in Fig. 7A and curves 4 and 5 in Fig. 7B progressively approach this ideal response, curve 5 being about the closest approach to the ideal which can be secured in practice. A radio expert would label curve 2 as "fair fidelity but very poor selectivity"; curve 3 would be designated as "poor fidelity and good selectivity" (this particular response curve might be satisfactory for radio-telegraphy work, where the side frequencies extend very little off resonance). In Fig. 7B, curve 4 represents an amplifier having good selectivity and fair fidelity, while curve 5 has excellent fidelity and good selectivity.

Resonance curves very similar to those just studied are used in portraying the abilities of R.F. amplifiers designed for television purposes. Remember, however, that the resonant frequencies here will be in the range extending from 40 to 90 megacycles, the band width will be 5 to 10 megacycles wide and stations will be separated as much as 15 megacycles. The same resonance curves can be used to portray the behavior of R.F. amplifiers which follow the modulated stages in transmitters.

Now we are ready to consider R.F. amplifier circuits in detail; you will learn how their desired features are obtained. A knowledge of how their operation is affected by changes in circuit parts will be of great value when you have to make repairs on incorrectly operating receivers.





FIG. 7B. Ratio resonance curves of R.F. amplitiers designed to have 10-kc. selectivity. Vertical scale gives values of RA.

SIMPLE TUNED R. F. AMPLIFIERS

Amplifier Loads. Almost any circuit which uses a vacuum tube having a control grid can be used as an R.F. amplifier provided that the proper plate load is used. If a simple resistor or coil is made to serve as the plate load, we have an untuned R.F. amplifier circuit which will amplify R.F. signals over a very wide band; this poor-selectivity characteristic is highly desirable in those cases where other radio devices are used for signal selecting and tuning purposes. If a reasonable amount of voltage amplification is to be secured, the ohmic resistance (or the reactance in the case of a coil) of the plate load (which is untuned) should be greater than the A.C. plate resistance of the tube.

Coil-Condenser Loads. You are already familiar with the fact that a coil and condenser connected in parallel, as in Fig. 8, behave exactly like a resistor of high ohmic value at the resonant frequency (the frequency at which the coil reactance exactly balances the condenser reactance), but offer a lower reactance at other frequencies. Changing the value of either

the coil or the condenser changes the resonant frequency. Thus this parallel coil-condenser resonant circuit is a very desirable load for an R.F. amplifier, giving high amplification of the desired signal, giving low amplification of other frequencies and permitting tuning to a desired frequency.

If coil L in Fig. 8 has an inductance of 250 microhenrys and a resistance R of 20 ohms, and condenser C is adjusted to 100 micro-microfarads (mmfd.), this circuit will act like a 125,000 ohm resistor at its resonant frequency (this can be proved by actual test or by computation with a formula given later in this book). Thus, when this circuit is used as a plate load, a high value of A.C. voltage is produced across terminals 1 and 2 at the resonant frequency, but the circuit acts either as a condenser or coil of low reactance to other frequencies (off-resonance frequencies), with a resulting low A.C. voltage for these frequencies; the circuit therefore possesses selectivity characteristics. Decreasing the inductance of the coil or decreasing the capacity of the condenser raises the resonant frequency of the circuit. Another advantage of this coil-condenser tuned load circuit is that it is not seriously affected by capacity between the electrodes of the tube; a high plate-to-cathode capacity simply means C can be smaller.



Coil-Condenser Load Connections. There are a number of different ways of connecting a coil-condenser resonant circuit into the plate circuit of a tube; four of the most important of these are shown in Fig. 9. Remember that a small resistance (that of the coil) is always present in a resonant circuit (it is usually neglected when preparing diagrams, in order to simplify the presentation of resonant circuits).

In Fig. 9A, resonant circuit L-C (often called a tank circuit) is connected directly into the plate circuit, with D.C. plate supply current flowing through coil L. The A.C. voltage which is developed across the tank circuit terminals is impressed upon the grid of the next R.F. amplifier stage through coupling condenser C_{κ} , which has negligible reactance to signal currents. This condenser also prevents D.C. plate current from flowing to the grid of VT_2 , the next tube. Resistor R_g has an important function, that of connecting the grid to ground in order that the C bias voltage developed by cathode circuit resistor R_c can act upon the grid of the tube. Obviously the ohmic value of R_s must be much higher than the resonant resistance of the tank circuit if it is not to affect the plate circuit. (Since $R_{\rm g}$ is in parallel with the plate circuit, a low value of R_s would absorb power, lowering the resonant circuit resistance and thus reducing the net plate circuit resistance; the result would be a reduction in the amplification of the amplifier and a broadening of the resonance curve. This broadening improves fidelity, and is therefore desirable in some cases.)

In Fig. 9B, choke coil CH is used in place of grid resistor $R_{\rm g}$. The reactance of this choke coil at any frequency to which the tank circuit may be tuned should be many times greater than the resonant resistance of the tank circuit, in order to secure a large A.C. voltage drop across the tank circuit.

In Fig. 9C the positions of the choke coil and the tank circuit are interchanged as compared with Fig. 9B; this arrangement, where the choke coil is in the plate circuit, is known as a shunt fed circuit, for the D.C. power is fed in parallel or in shunt with the L-C circuit. When the resistance of R_s in Fig. 9A and the reactance of CH in Figs. 9B and 9C are many times greater than the resonant resistance of the tank circuit, the values of L and C in the tank circuit control the R.F. signal output; the resonant circuit thus becomes the plate load. The following important facts apply to all three circuits:

1. The values of L and C determine the resonant frequency of the tank circuit. For practical reasons it is customary to tune the condenser; decreasing the capacity of the condenser increases the resonant frequency of the tank circuit. Coil L is still shunted by a capacity, however, when condenser C is set at minimum capacity; this residual capacity consists of: a, the distributed capacity of coil L; b, the minimum capacity of the condenser; c, capacities between the tube electrodes and between the leads in the circuit. In a well designed practical circuit the minimum or residual tank capacity is about one-ninth of the maximum tank capacity; this means that the maximum frequency to which the tank circuit can be tuned is about 3 times the lowest frequency. An example of this is the average broadcast band receiver, which tunes from about 500 to 1,500 kc, with a single set of coils.

2. The effective resistance of the tank circuit at resonance is determined by the inductance of the coil, the capacity of the condenser and the inherent tank circuit resistance.

3. If the tank circuit at resonance has a coil of high inductance and a condenser of low capacity, the resonant resistance will be high. Radio men say that a tank circuit with large L and low C has a high L/C ratio. The lower the inherent resistance of the tank circuit, the greater will the resistance of a tank circuit be at resonance.

4. In a single-tube tuned R.F. amplifier stage which uses a parallel resonant circuit as a plate load, the maximum possible amplification is the mu of the tube, and this value is obtained when the resonant resistance of the plate load is many times greater than the A.C. plate resistance of the tube. Triode tubes have relatively low A.C. plate resistances and hence this condition is quite easy to secure, but with tetrode and pentode tubes the load ordinarily cannot be made high enough in value to give maximum amplification.

5. When the tank circuit is tuned to higher frequencies in the usual manner hy reducing the value of capacity C, the L/C ratio increases, raising the resonant resistance of the load.

Tuned Transformer Circuit. In Fig. 9D is perhaps the most widely used R.F. amplifier circuit, the so-called tuned transformer circuit. The power required for tank circuit L-C must be supplied to the primary coil L_p by the tube. When the L-C tank circuit acts like a resistor, as at resonance, the reflecting properties of the transformer make coil L_p act like a resistor (the

* There are two formulas for securing the resonant resistance of a tank circuit like that in Fig. 8; the radio engineer uses whichever is more convenient. The L $\omega^2 L^2$ formulas are: $R_{\rm R} = \frac{L}{RC}$, and $R_{\rm R} = \frac{\omega^2 L^2}{R}$, where $R_{\rm R}$ is the resonant resistance of the tank-circuit in ohms, R is the inherent tank circuit resistance in ohms, ω is 6.28 times the frequency in cycles, L is the inductance of the coil in henrys and C is the capacity of the condenser in farads.

leakage inductance of primary L_{\bullet} affects this condition, but this inductance is so small that it can be neglected). Facts 1, 2, 3 and 5 brought out in connection with the first three circuits in Fig. 9 hold true also for this circuit, and the following additional facts apply to this circuit only:

1. Increasing the number of turns on primary winding $L_{\rm P}$ or increasing the coupling between $L_{\rm P}$ and L raises the reflected resonant resistance appearing between terminals I and 2. We are actually increasing the mutual inductance M of the coils when we add turns or move the coils closer together; thus we can say that increasing the mutual inductance increases the effective resonant resistance of the plate load.



FIG. 9. Four ways of connecting the resonant L-C load circuit into an R.F. amplifier stage so it will act as a plate load. These circuits apply also to screen grid and pentode tubes; triodes are shown here merely for simplicity.

2. When a tuned transformer load is used, maximum voltage amplification is obtained when the resonant resistance of the tuned load (as it appears in the plate circuit) is equal to the A.C. plate resistance of the tube, for the tank circuit then absorbs the greatest amount of power.

3. Circuit L-C in Fig. 9D is a series resonant circuit (because the voltage induced in it acts in series with L and C), and hence acts as a coil when tuned below resonance and as a condenser when tuned above resonance. The effects at terminals I and 2 are opposite, however, because of transformer action.

HOW TUBE CAPACITIES AFFECT R. F. AMPLIFIER CIRCUITS

Inter-Electrode Capacities. When we study an ordinary triode tube in detail, we realize that the grid, plate and cathode are tiny condenser plates which introduce into the tube circuits the three capacities shown by dotted lines in Fig. 10. These inter-electrode capacities are: 1, the grid-to-cathode capacity, $C_{\rm gk}$; 2, the plate-to-cathode capacity, $C_{\rm pk}$; 3, the grid-to-plate capacity, $C_{\rm gp}$. It is the grid-to-plate capacity, $C_{\rm gp}$, which prevents the grid and plate circuits from being entirely independent of each other.

In the R.F. amplifier circuit in Fig. 10, the grid-to-cathode capacity $C_{g^{k}}$ acts simply as if it were an extra condenser connected in parallel with C_{1} ; the plate-to-cathode capacity $C_{g^{k}}$ acts as a condenser connected in parallel

with C_2 . By-pass condensers C_c and C_p in Fig. 10 are of such high capacity that they act as low reactance paths for A.C. currents, and therefore need not be considered in this discussion.

In Fig. 11 the R.F. amplifier circuit of Fig. 10 has been modified to include only those parts which affect the tube as an amplifier; $C_{\rm gk}$ and $C_{\rm pk}$ have been placed in their effective positions. Capacity $C_{\rm gp}$ can, according to a complicated mathematical analysis which need not be taken up here, be considered as equivalent to an extra resistor $R_{\rm R}$ and an extra condenser $C_{\rm R}$, both in parallel with the grid tank circuit C_{1} - L_{1} . Remember that $R_{\rm R}$ and $C_{\rm R}$ are present as effects only, not as actual devices; their exact equivalent values vary greatly with circuit conditions, being dependent upon the capacity of $C_{\rm gp}$, upon the resonant frequency and nature (resistive, inductive or capacitive) of the plate circuit load, and upon the over-all amplification of the stage.

Coil L_2 in Fig. 11 acts with C_2 and C_{pk} to form a parallel resonant tank sircuit which is the equivalent of a high resistance at resonance. When this parallel resonant circuit is tuned below resonance (to a frequency *lower*)



FIG. 10. The three inter-electrode espacities which enter into the operation of an R.F. smplifier are here indicated by dotted lines.





than that of the incoming signal), the tank circuit will act as if it were a condenser. (This is because decreasing the resonant frequency by increasing the capacity of C_2 lowers the reactance of the condenser but does not affect the reactance of the coil; in a parallel circuit such as this it is the lowest reactance [the reactance of L_2 or C_2] which governs the nature of the tank circuit off resonance.) Likewise, when this parallel resonant circuit is tuned above resonance (to a frequency higher than that of the incoming signal), the tank circuit acts as a coil.*

^{*} When a resonant load circuit is loosely coupled inductively to the plate circuit of a tube, as is done in Fig. 9D, the tuned secondary (L-C) is purely resistive when tuned to resonance. Because of the leakage inductance of primary L_p , however, the effect of the load in the plate circuit (as measured between points 1 and 2) will be *inductive*. When the L-C circuit is tuned off resonance, the reactance which it reflects into primary L_p will always be opposite in nature to the reactance which the L-C circuit itself appears to have. Thus, when series resonant circuit L-C is tuned above resonance it acts as a capacity and is reflected as an inductance; when tuned below resonance it acts as an inductance and reflects as a capacitance. In order to make the effect of the load purely resistive across terminals 1 and 2, then, it is necessary to tune the load slightly below resonance, so it reflects a capacitance which will balance out the leakage inductance of the primary; in doing this we are really using the secondary L-C circuit to tune the primary L_p (which is in effect the load) to resonance.

The circuit in Fig. 11 possesses a number of interesting and important characteristics, some of which are portrayed by the curve in Fig. 12. These facts are listed here, for they mean a lot to any one who will work with R.F. amplifiers. The following facts apply equally well to the circuits in Figs. 9 and 10 (if points 1 and 2 in Fig. 9D are considered as the load terminals).

1. When the load is tuned exactly to the frequency of the incoming signal, the capacity of C_R is a maximum and is equal to the capacity of C_{ep} plus a value equal to the capacity of C_{ep} multiplied by the true amplification of the stage. When the load is tuned either above or below the signal frequency, the true amplification of the stage is greatly reduced and the capacity of C_R therefore decreases rapidly, as is shown in Fig. 12.

2. When the load is tuned exactly to the incoming signal, the equivalent ohmic value of R_R is so high that it can be neglected.

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3. When the load acts as a condenser, the equivalent ohmic value of R_R is greatly reduced. The selectivity response curve of the amplifier is consequently broadened and the amplification of the stage is reduced.

4. When the load acts as a coil, the tube is actually feeding power back into the grid circuit, thus increasing the amplification of the stage and improving selectivity. For convenience this is often explained by saying that the ohmic value of R_u becomes *negative*: a negative resistance in shunt with a real resistance will actually increase the net resonant resistance of $L_1 \cdot C_1$, clearly indicating that the circuit is capable of producing a greater resonant voltage.

5. If the load acts as a coil of sufficiently high reactance, the tube will feed sufficient energy back to the grid to make the circuit oscillate at a frequency determined essentially by the values of L_1 , C_1 , $C_{\rm gs}$ and $C_{\rm R}$.

Practical Considerations. The curve in Fig. 12 also contains the answers to some very practical questions concerning the operation of amplifiers in general, including audio or video amplifiers. Keep the following facts in mind when you work with amplifiers and you will have no difficulty in finding causes and cures for annoying squeals or for poor fidelity.

1. In an audio or picture frequency amplifier. a high plate load resistance or reactance will give high amplification and make C_R very high in capacity. The higher frequencies in the audio or picture signal will therefore be attenuated (by-passed to ground before they reach the grid). A low-impedance load gives less attenuation.

2. If high plate load impedance in an audio or picture frequency amplifier stage is accompanied by an inductive plate load condition, undesirable oscillation may take place.

3. In an R.F. amplifier using a triode tube, any stage having plate and grid circuits arranged as in Fig. 10 will go into oscillation whenever the plate tank wircuit is made sufficiently *inductive* by tuning it above the signal frequency. Two R.F. signals, the incoming signal and the oscillating frequency, will then reach the grid and be amplified by the tube; these two frequencies will be mixed together by the detector stage, and if the difference in their frequencies is in the audio range, an annoying squeal will be heard in the loudspeaker.

4. Oscillations can take place even without an incoming signal if the resonant frequency of L_2 - C_3 , the plate load, is considerably higher than the resonant frequency of L_1 - C_1 in the grid circuit, for under this condition the plate tuned circuit will be *inductive* at the resonant frequency of the grid tuned circuit. These being R.F. oscillations, they will not ordinarily be heard in the loudspeaker. If, however, this condition for R.F. oscillation exists in two different stages, two R.F. signals may go through the system and beat with each other at the detector, producing an A.F. note which will be heard as a squeal. This explains why some receivers produce squeals even when not tuned to a station; the remedy simply involves adjusting one or more of the tuned circuits.



GETTING RID OF FEED-BACK

When triode tubes are used in amplifier circuits, it is the grid-to-plate capacity which is of vital importance; the other two inter-electrode capacities, $C_{\rm gk}$ and $C_{\rm pk}$, simply reduce the tuning range of the amplifier slightly at the highest frequencies.

Grid Suppressors. The oscillation or regeneration caused by the grid-toplate tube capacity in a triode tube is an especially serious problem in single dial receivers, for these have no provisions for tuning out an annoying squeal at a particular setting of the tuning dial. Since only triode tubes were available in the early days of radio, engineers were compelled to develop a number of solutions for this regeneration or feed-back problem. One of these involved loading the grid tank circuit $(L_1-C_1 \text{ in Fig. } 10)$ with a resistor which made the resonant resistance of the grid tank circuit less than the ohmic value of the equivalent negative resistance R_R . (Engineers have proved that R_R , in addition to being negative must be lower in ohmic value than the resonant resistance of the grid tank circuit before oscillation can occur.) Another feed-back-killing scheme involved inserting a re-

FIG. 12. This curve shows how the equivalent capacity CR of the circuit in Fig. 11 varies when the plate tank circuit of an R.F. amplifier is tuned to, above and below the frequency of an incoming signal.



sistor in the grid lead, as at point 1 in Fig. 10; the resistor was in this case known as a grid suppressor. Lowering the amplification of the stage, such as by using a low L/C ratio in the plate tank circuit, was another way of preventing regeneration.

Neutrodyne Circuits. In the early days of radio, neutrodyne receivers were "all the rage"; these used an ingenious method of preventing regeneration while retaining all of the amplification of a stage. An A.C. voltage taken from the plate circuit of the R.F. amplifier was fed back into the grid circuit out of phase to balance out, buck out or *neutralize* the undesirable feed-back voltage caused by grid-to-plate capacity.

Figure 13A is an example of one of these early neutrodyne circuits; regeneration occurred here because the A.C. voltage across coil L_1 fed back to the grid tank circuit through inter-electrode capacity $C_{\rm gp}$. To offset this, coil L_2 was coupled with and connected to coil L_1 in such a way that the voltage which it fed through neutralizing condenser $C_{\rm N}$ to the grid was out of phase with the undesired feed-back voltage; the two voltages therefore bucked each other and, when $C_{\rm N}$ was adjusted to make the two voltages equal, they exactly cancelled each other and eliminated regeneration effects.

Figure 13B gives another neutrodyne circuit; no additional coil winding was needed here, for the required out-of-phase A.C. voltage was secured

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by tapping the secondary winding of the R.F. transformer. There were many variations of these two neutrodyne circuits, but all were essentially the same in their operation. Neutrodyne circuits were not entirely practical, for they were easily thrown out of adjustment. Any changes in the circuit parts, or the installation of a new tube, made it necessary to readjust neutralizing condenser $C_{\rm N}$.

Engineers realized that the grid-to-plate capacity in the tube was the real cause of all their regeneration troubles, and finally called in vacuum tube engineers to design a tube which had negligible grid-to-plate capacity. The screen grid or shielded grid tube, which you know as the tetrode, was the result; we will consider this tube next.

THE SCREEN GRID TUBE

As you know, it is current flowing from the plate through inter-electrode capacity C_{gp} to the grid which results in regeneration in amplifier circuits. If an additional grid is placed between the control grid and plate of a triode, and this new grid, which we call the *screen grid*, is connected to









ground, the feed-back current will flow to the screen grid and then to ground, and will not reach the control grid at all. Tube engineers also realized that this new grid could be made to aid the plate in pulling electrons out of the space cloud near the cathode. The resulting circuit arrangement, in which the screen grid is at zero potential for A.C. and at a high positive potential with respect to the cathode for D.C., can be seen in Fig. 14A. The screen grid is connected directly to one terminal of the power supply, and the plate is connected to a higher voltage terminal (B++), while by-pass condenser C_2 is placed between screen grid and ground to provide a path to ground for the A.C. feed-back currents. The capacity of C_2 must be much higher than the capacity existing between screen grid and control grid in order to make the feed-back current take the desired path to ground.

Now you can readily see that if the screen grid by-pass condenser in an R.F. amplifier (such as C_2 in Fig. 14A) should open, as it often does in radio equipment, the feed-back current will flow past the screen grid to the control grid, and oscillation will occur, resulting in a squeal in the loud speaker output. Replacement of the screen grid by-pass condenser (C_2) will eliminate the squeals.

The effect of the screen grid in reducing the grid-plate capacity of a tube can be determined from any tube chart. A comparison of the inter-electrode capacity values for a typical triode and a typical tetrode tube is given below:

> Type 27 (triode) $-C_{gp} = 3.3 \text{ mmfd.}; C_{gk} = 3.1 \text{ mmfd.}; C_{pk} = 2.3 \text{ mmfd.}$ Type 24A (tetrode) $-C_{gp} = .007 \text{ mmfd.}; C_{gk} = 5.3 \text{ mmfd.}; C_{pk} = 10.5 \text{ mmfd.}$

Observe that in the tetrode the introduction of the screen grid has reduced the grid-to-plate capacity about 470 times, an amount which is more than sufficient to prevent regenerative feed-back. $C_{\rm gp}$ cannot be entirely eliminated, however, for there is always a little leakage current from plate to grid. Surrounding the plate completely with the screen grid eliminates a great deal of this leakage, at the same time increasing the plate-tocathode capacity, but increasing $C_{\rm pk}$ does not appreciably affect the performance of a tuned R.F. amplifier; it simply means that less capacity will be needed in the plate circuit variable condenser, and a reduction in the highest frequency to which the T.R.F. amplifier can be tuned.

FIG. 14A. R.F. amplifier circuit using a screen grid tube. In such screen grid tubes as the 24A, the screen grid completely surrounds the plate, as indicated by the light dotted lines, but the schematic disgram for a screen grid tube generally shows only the heavy lines here indicated.



With the troublesome grid-to-plate capacity out of the way, radio tube engineers proceeded to build greater amplification into the screen grid tube by moving the control grid closer to the space cloud and moving the plate farther away from the cathode. While this did not materially increase the electron flow to the plate, it did increase the A.C. plate resistance. The screen grid tube therefore has a high amplification factor, a high A.C. plate resistance and an average value of mutual conductance; the following table gives a comparison of these values, together with rated plate current values for a triode and a tetrode:

Type 27 (triode): $\mu = 9$; $g_m = 1,000$ micromhos; $r_p = 9,000$ ohms; $l_p = 5.0$ ma. Type 24A (tetrode): $\mu = 400$; $g_m = 1,000$ micromhos; $r_p = 400,000$ ohms; $l_p = 4.0$ ma.

Details of construction of a typical screen grid tube, the type 32, are shown in Fig. 14B; note how completely the plate is surrounded by the screen grid.

The Performance of a Screen Grid Tube. The behavior of a tetrode or screen grid tube as an amplifier is expressed by the family of E_p - I_p curves given in Fig. 15; these curves give average characteristics for the RCA Radiotron type 24A screen grid tube. Notice that plate current is referred to as I_b instead of I_p . In securing data for these curves, the tube was operated at a normal filament voltage of 2.5 volts (either A.C. or D.C.) and normal screen grid voltage of 90 volts D.C., and the plate voltage was

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varied from zero to a maximum for each value of D.C. control grid voltage (for 0, -1.5, -3.0, -4.5 and -6.0 volts).

For any one value of control grid voltage, plate current rises quite rapidly at first, as the plate voltage is gradually increased from zero. The screen grid has such tremendous electron-pulling power, however, that electrons <u>approach it at very high speeds</u>, pass right through its widely <u>spaced wires</u> and hit the plate with terrific impact, knocking electrons out of the plate; this effect is known as *secondary emission*. Some of these secondary electrons return to the plate, but a great number of them are attracted to the screen grid, which also is at a high positive D.C. voltage.



Courtesy Hygrade Sylvania Corp.

Fig. 14B. Construction of a typical screen grid tube, a type 32 tetrode. The control grid is at A. The inner part of the screen grid (mounted between the plate and the control grid) is at B, while the euter part of the screen grid (which surroonds plate C) is at D. At the right is the completed tube.

Curve I_{c2} , representing screen grid current, shows you that for low plate voltages the screen grid current is actually higher than the plate current, indicating that the screen grid collects a high proportion of the electrons which "bounce" off the plate and also attracts electrons directly (the screen grid is at a higher potential than the plate in the region to the left of line AA). It is for these reasons that in the region between zero and 90 volts (labeled "VALUES UNSTABLE" in Fig. 15), increases in plate voltage actually cause the plate current to go down. For plate voltages above 90 volts, plate current rises and screen grid current decreases gradually to a very low value.

When the plate tank circuit in an R.F. amplifier contributes its resonantproduced A.C. voltage to the D.C. plate voltage, it makes the plate-tocathode voltage swing from high to low values. In order to show how plate current varies under this condition, we can use the load line in Fig. 15; this load line gives us plate current and plate voltage values directly, making it unnecessary to compute the voltage drop across the load.

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The plate voltage of the screen grid tube in Fig. 15 cannot be allowed to swing below the screen grid voltage, (90 volts in this case) if distortion is to be prevented. Incidentally, if the plate voltage in an R.F. amplifier circuit like Fig. 14A should swing down to about 40 volts, the circuit will act as an oscillator instead of an amplifier; under this condition the screen and plate possess a negative resistance characteristic and supply power to the plate tank circuit. A screen grid tube gives better results than a triode in an R.F. amplifier provided that we limit the plate voltage swing to the linear portion of the E_g - I_p characteristic curve.

THE SUPPRESSOR GRID TUBE

Tube designers quickly realized that some means of forcing secondary electrons to return to the plate in a screen grid tube instead of going to the screen grid would greatly improve the amplifying characteristics of the tube at low plate voltage. One scheme for accomplishing this, actually tried in the type 48 power tetrode tube, involved attaching thin metal fins



around the inside of the plate electrode, as shown in Fig. 16A. These fins exert control over the secondary electrons for a considerable distance after they have bounced off the plate. The fins attract far more secondary electrons than does the screen grid, and these electrons then travel back along the fins to the plate. The chief objection to this method is that the fins are expensive to construct; electrical means for accomplishing the same results have been developed.

The introduction of an extra grid between the plate and the inner section of the screen grid proved to be the electrical answer to the problem of driving secondary electrons back to the plate. This extra grid is known as the suppressor grid, for it actually *suppresses* the effects of secondary emission.

Here is how a suppressor grid acts: First of all, it is connected to the cathode or to a terminal which is more negative than the cathode. When an electron bounces off the plate and tries to flow to the positively charged screen grid, it is repelled just enough by the suppressor grid to be forced back to the plate. Even when the suppressor grid is at zero or cathode potential, its action is sufficient to make secondary electrons prefer the plate to the screen grid. Why is it that the suppressor grid has very little effect upon the electrons flowing from the cathode to the plate, but a great deal of control over those which reverse their direction at the plate? Speed is the answer; an electron moving from the cathode through the control grid and the screen grid to the plate has developed so much speed by the time it reaches the suppressor grid that it goes right through the coarse wire mesh without slowing up at all, but secondary electrons coming from the plate have very little speed. We might compare the suppressor grid to a thin sheet of steel, which is easily pierced by a bullet from a high powered gun but which is able to stop a slow speed bullet from a small rifle. This one-way action of the suppressor grid is so efficient that it is even possible to operate the screen grid at the same potential as the plate, improving the electron flow to the plate without affecting secondary emission.

Figure 16B shows a typical circuit for a suppressor grid or pentode tube. Notice that the suppressor grid is connected directly to the cathode. The screen grid is often connected to the same power supply terminal as the plate; thus terminals 1 and 2 may be connected together in some circuits, as they usually are when power pentode (suppressor grid) tubes are used.



The quality of a radio tube is judged entirely by its characteristics. A family of E_p - I_p characteristic curves for a pentode tube designed for R.F. amplification is given in Fig. 17; notice that the suppressor grid has eliminated the unstable region below the screen grid voltage value which was present in the screen grid tube characteristic curves in Fig. 15. The plate current of this pentode tube practically reaches its full value at low plate voltages, indicating that secondary emission is not depriving the plate of its current. A tube with characteristics like this can be made to swing over wide plate current and plate voltage values, giving maximum amplification and maximum power output for the tube.

VARIABLE-MU OR SUPER-CONTROL TUBES

You may already have heard of variable-mu tubes, also called supercontrol tubes, for they are quite commonly used in amplifier circuits where the gain (the amplification) is varied by varying the C bias voltage of the tube. These tubes solve the problem of controlling volume by providing a mutual conductance (and therefore an amplification factor μ) which can be varied by changing the setting of the C bias voltage. But let us first see what happens when we try to control volume in this way with ordinary tubes. Cross-Modulation. When two stations are heard at the same time on a radio receiver, one being a weak station to which the receiver is tuned and the other being a powerful local station, and this is observed for a number of other desired weak stations as well, cross-modulation exists.



Here is what happens: The weak signal is amplified in the normal manner by the first R.F. amplifier stage, but because of poor selectivity the strong signal also reaches the amplifier tube and is demodulated because it swings the grid of the first tube beyond the linear region of the E_{z} - I_{p} characteristic curve. Plate current thus varies at the modulation frequencies of the strong station, just as in any detector, and these A.F. variations serve to modulate the desired R.F. carrier. This carrier, now having both the desired and the undesired modulation signals, goes through the receiver in



FIG. 18A. Characteristic curves for a type 57 or 6C6 pentode tube having both screen and suppressor grids.



the normal manner and both modulation signals are demodulated together.

The characteristic curves for a typical R.F. amplifier tube are given in Fig. 18A; let us consider an actual example of cross-modulation, where the tube represented by these curves is placed in the R.F. amplifier circuit of Fig. 18B. A normal C bias of -3 volts can be obtained for this tube by adjusting P_1 . If the desired signal has a peak voltage of one-half volt, it will swing the grid over a linear portion of the grid voltage-plate current characteristic curve (I_b) and the signal will be amplified in a normal man-

ner. If any undesired signal greater than 3 volts gets into this stage, however, it will swing the grid down beyond the bend in the I_b curve or even beyond the plate current cut-off point. The resulting rectification of the undesired signal causes cross-modulation.

Modulation Distortion. The E_{g} - I_{p} curve (curve I_{b} in Fig. 18A) for an ordinary screen grid tube has a very pronounced bend in the region of cutoff; the result is that signal voltages which swing the control grid more positive cause large changes in plate current, whereas signal voltages which swing the grid more negative give only very small changes in plate current. The harmonics produced under this condition can be partially filtered out by resonant circuits, but the more serious effect, distortion of the wave form of the modulation accompanying the carrier, is difficult to remedy; modulation distortion is the term applied to this operating defect of a screen grid tube.

In order to eliminate cross-modulation and modulation distortion effects yet still control volume by means of the C bias voltage, we need a tube which will reduce its mutual conductance as the C bias is made more negative, but which will not have a *sharp* bend in its E_g - I_p curve at low current values; we need a tube which requires a very high negative C bias to produce plate current cut-off.

An ordinary screen grid tetrode or pentode tube has a sharp plate current cut-off and, since the control grid wires are close together and near the cathode, only a small negative C bias is needed to drive the plate current to zero. This construction, as you know, gives a high amplification factor. If, now, we modify the construction so that some of the control grid wires are farther away from the cathode than others or so some of the control grid wires are closer together than others, we get a tube which has the desired characteristics. The variable spacing scheme is illustrated in Fig. 19A; this is the arrangement commonly used for variable-mu tubes, since it simplifies construction problems.

This variable spacing of control grid wires gives to the tube two distinct characteristics, one for each spacing, as you would expect. At small negative C bias voltages the action of the tube is essentially the same as if all grid wires were uniformly spaced at the maximum distance, but at high negative bias values the closely wound turns block electron flow; electrons continue to pass through the widely spaced grid turns to the plate until the highly negative cut-off bias is reached, and thus we obtain the desired remote cut-off characteristic.

The mutual conductance and plate current curves for a representative variable-mu tube are given in Fig. 19B, with the plate current curve $(E_g - I_p \text{ curve})$ for an ordinary screen grid tube (the type 57) shown for comparison. If you compare the plate current cut-off points and the bends in the two I_p curves, you can verify the statements just made concerning variable-mu tubes.

Ordinarily screen grid and pentode tubes are quite satisfactory for those circuits where the C bias is such that the A.C. signal swings over a linear part of the E_{g} - I_{p} characteristic, but only a very limited variation in g_{m} is permissible if cross-modulation and modulation distortion are to be

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eliminated. Variable-mu screen grid tubes and variable-mu pentodes (often called super-control pentodes) permit considerable variation in their values of g_m ; these tubes are well suited for those stages of a well designed amplifier where a C bias type of gain control is used, for they will reduce cross-modulation and modulation distortion.

PRACTICAL R. F. PENTODE CIRCUITS

Amplifier circuits containing ordinary pentode tubes or super-control pentode (variable-mu) tubes are basically the same as the triode amplifier circuits shown in Fig. 9. Remember, however, that pentode tubes have a very high A.C. plate resistance; this means that the conditions for maximum amplification (a load resistance many times higher than the A.C. plate resistance if an ordinary resonant circuit is used as load, or a load resistance equal to the A.C. plate resistance in the case of a tuned trans-



FIG. 19A. This sketch shows the variable spacing used for the control grid wires in a variable mu tube.

FIG. 19B. Characteristic curves for a variable mu pentode tube (type 58), with plate current curves for ordinary pentode (type 57) shown for comparison.

former load for a single-tube stage) are seldom if ever secured. A tank circuit load which is designed for a triode tube is purposely given a low resonant resistance, so obviously a triode load will not be satisfactory in a pentode circuit.

A two-stage R.F. amplifier using pentode tubes and containing a number of different types of R.F. circuits is shown in Fig. 20. The first tube is an ordinary pentode; a 1,500 ohm cathode-to-ground resistor supplies it with an essentially constant automatic C bias. Condensers marked .1 (.1 mfd.) serve as low-reactance paths for A.C. currents. The second R.F. amplifier stage, containing a super-control pentode tube, is coupled to the first stage by condenser C, which is used both as a D.C. blocking condenser and as an A.C. coupling condenser. The capacity of C is quite low, being about 10 mmfd.; this capacity generally is secured simply by twisting two insulated wire leads together once or twice.*

^{*} When the R.F. choke and the tank coil are constructed as a single unit (but with no inductive coupling between them), a stiff wire is sometimes connected to the upper end of choke RFC and looped partly around the grid end of the tank coil L_2 to provide the required low coupling capacity C.

The type 58 super-control pentode tube feeds into its load through a transformer having a tuned primary. The gain of this R.F. amplifier is controlled by varying the negative C bias which is applied to the type 58 tube; potentiometer P in the voltage divider network provides a means of varying this C bias. Choke coil RFC_1 in the plate supply lead of the super-control pentode prevents A.C. signal currents from entering the power supply circuit.

Pentode R.F. amplifiers of the type just described are used for all frequencies. With R.F. values between 100 kc. and 20 megacycles it is simply necessary to design the resonant circuit to respond to the desired frequency and to have a resonant resistance which gives sufficiently high amplification for desired results. By-pass condensers and choke coils must, of course, be of the proper values to keep A.C. signals in their proper paths. At the ultra-high frequencies, however, such as those used in television systems, the grid-to-cathode, grid-to-plate and plate-to-cathode capacities in the tubes become appreciable in value, especially in stages which are designed for high voltage gain; in these cases acorn-type tubes, which have no bases and are so designed that leakage and inter-electrode capacities are a minimum, are used.



FIG. 20. Circuit of a two-stage R.F. amplifier using pentode tubes. The small circle on a condenser symbol indicates that the condenser is adjustable. Numerals alongside a condenser represent its capacity in microfarads. All .1 mfd. condensers are by-pass condensers.

BAND-PASS R. F. CIRCUITS

All of the R.F. circuits which you have just studied have had single peak frequency response curves like that shown in Fig. 7A; these are, of course, perfectly satisfactory where the side-band frequencies involved are within narrow limits. In order to approach the ideal flat response curve shown in Fig. 7B, multiple resonant circuits are required. It is possible to flatten the curve produced by a single tuned circuit if the tank circuit is shunted by a resistance, but this reduces the gain considerably and simply gives an approach to the ideal response.

When a number of tuned circuits are used in cascade, as they are in Fig. 20, it is possible to secure an approach to an ideal response curve by tuning one or more circuits off the incoming signal frequency. For example, circuit 2 could be tuned to the desired 1,000 kc. signal, circuit 1 could be tuned to about 995 kc., and circuit 3 could be tuned to about 1,005 kc. The over-all

response curve is the product of the responses of each stage. By careful adjustment, very nearly ideal response, with three more or less pronounced peaks, like those shown in Fig. 21, can be obtained. Considerable gain is of course sacrificed in order to secure this response; another disadvantage is that two tubes are required to provide the three tuned circuits.

It is possible, however, to obtain flat response with only a single R.F. stage; several such stages can be used to give the desired amount of amplification, for the gain of the stages is cumulative. Double and sometimes triple tuned circuits are used in place of single resonant circuits; a typical double tuned circuit which is widely used in the I.F. amplifiers of superheterodynes is shown in Fig. 22. Circuits L_1 - C_1 and L_2 - C_2 are usually identical circuits, coupled to each other by mutual inductance M. When both resonant circuits are tuned to give maximum output, a single peak response curve is obtained; if each tuned circuit has low circuit resistance, the frequency response curve will be pointed or peaked, whereas high circuit resistance gives a broad, rounded curve. When circuit L_1 - C_1 is tuned to a frequency above the incoming signal (by reducing the capacity of C_1) and circuit L_2 - C_2 is tuned below the incoming signal (by increasing the capacity of C_2), the resulting resonant curve will have a double hump. The



primary could just as well be tuned below resonance and the secondary above resonance for band-pass results.

The use of three resonant circuits, each coupled to the next and each adjusted to a different frequency, as the load of an R.F. amplifier gives a triple hump response curve. Of course, each extra coil wastes a certain amount of power, lowering the effective value of plate load resistance and consequently reducing the gain, but this loss in gain can be compensated for by using several such R.F. stages in cascade. A tuned amplifier having two or three tuned circuits obviously cannot give as much amplification, even when each circuit is tuned to resonance, as a single tuned circuit because of the losses in the extra coils. Single tuned circuits are therefore often referred to as high gain R.F. amplifiers.

Figure 23 shows another way of using two resonant circuits to give a double peak response curve; capacity C here couples the two circuits together, for the voltage developed across C by the L_1 - C_1 circuit becomes the source voltage for circuit L_2 - C_2 . Coil L_1 is tuned by condensers C_1 and C in series, while coil L_2 is tuned by condensers C_2 and C in series. Since condenser C is usually fifty to one-hundred times as large as either C_1 or C_2 . its tuning effect is negligible.

Condenser C can be replaced with a coil if desired, giving an inductively coupled circuit; the inductance of coils L_1 and L_2 should then be fifty to one-hundred times that of the coupling coil. The band width of a double tuned coupled circuit like this is roughly dependent upon the reactance of the coupling device; this reactance increases with frequency when inductive coupling is used, but decreases with frequency when capacitive coupling is used. Thus, in Fig. 22, where inductive coupling is used, the frequency response curve will be broadest when the circuit is tuned to the higher frequencies, while for the circuit in Fig. 23, which has capacitive coupling, the frequency response curve will be broadest at the lower frequencies to which the circuit is tuned. Quite often both inductive and capacitive coupling are used in order to secure a constant band width, especially when a circuit is to be tuned over a wide range of frequencies.

The question of band width is highly important in high fidelity sound receivers and especially in television receivers. In high fidelity sound receivers the band width is generally between 10 and 20 kc., while in television receivers the band width may be as much as 6,000 kc. The greater the band width of an amplifier, the less attenuation there is of the higher side frequencies, and the better is the fidelity of response.



FIG. 23. The resonant circuits here couple the antenna circuit te the grid circuit of the first R.F. amplifier stage, giving a doublepeak response characteristic. r

FIXED R. F. AMPLIFIERS

At one time in the development of the radio industry, the tuned radio frequency amplifier patents were available only to a few radio manufacturers. This compelled other manufacturers to develop fixed R.F. amplifiers in order to get around these patents; tuning was accomplished with a resonant circuit placed ahead of the amplifier. In fixed R.F. amplifiers it was possible to use resistance-capacity coupled circuits, especially with triode tubes which had a low A.C. plate resistance; if screen grid tubes were used, however, a high plate load resistance and extremely high plate supply voltage were required.

A more practical solution to the problem of designing a fixed R.F. amplifier was found in the untuned air core transformer having a primary and secondary which were tightly coupled together; in fact, primary and secondary were wound simultaneously on a single coil form. The resulting circuit, shown in Fig. 24, gave excellent results with triode tubes. The coupling between the primary and secondary windings in the transformer was sometimes made even closer by using pulverized iron cores. Fixed R.F. amplifiers are seldom used today, as patents on tuned R.F. amplifier circuits, which give much better results, are available to practically all receiver manufacturers.

POWER R. F. AMPLIFIERS

Maximum power amplification in any single-tube R.F. amplifier stage is obtained when the A.C. plate resistance of the tube is equal to the resonant resistance of the plate tank circuit. When maximum power is desired in the tank circuit, the tubes should have a high mutual conductance. In power R.F. amplifiers, especially those used in transmitters, the circuit must be designed to make maximum use of plate power; since it is ordinarily not feasible to design tank circuits which will have resonant resistances comparable to the A.C. plate resistances of ordinary tetrode and pentode tubes, special power R.F. amplifier tubes have been designed, these having a low A.C. plate resistance which is secured by reducing the amplification factor of the tube and by raising the cathode emission. The use of special tubes is the essential difference between power R.F. amplifiers used in transmitters and those used in receivers. Some power amplifier circuits are designed for peak response at a single frequency, while others (such as circuits handling modulated R.F. currents) are designed for band-pass operation.

Efficiency. In high power R.F. amplifiers such as are used in transmitters, the A.C. power in the tank circuits may be hundreds or even thousands of watts; in cases like these the efficiency of operation of the tube is of vital importance.* When



FIG. 24. A fixed R.F. amplifier circoit, with tuning being accomplished by the three resonant circuits which precede the first triode tube.

the C bias of an R. F. amplifier is set to make the input signal swing over a linear portion of the dynamic $E_{\rm g}$ - $I_{\rm p}$ curve, so the tube is operating as a *class A amplifier*, an efficiency of about 30 per cent is obtained even if the grid is driven positive for a part of each cycle.

The plate resistance of the tube is to blame for the greatest part of this wasted power, for all plate current must flow through this plate resistance. Greater efficiency can be obtained by cutting down the time during which plate current flows; by setting the C bias for plate current cut-off, no plate current will flow at normal plate voltage when there is no excitation, and plate current will flow only for one-half of each cycle when grid excitation is applied; this condition, where the tube is operating as a class B amplifier, is shown graphically in Fig. 25. Efficiencies of 50% are possible with this circuit. Observe that although the grid excitation voltage e_g is a perfect sine wave in form, the plate current i_p is a half sine wave; thus power is being supplied to the plate load circuit for only half of each cycle.

Engineers have proved, both by calculation and experiment, that the half sine wave plate current i_p in Fig. 25 is made up of a direct current component and an A.C. component having *even* (second, fourth, sixth, eighth, etc.) harmonic components; all these components are trying to flow through the L-C plate tank circuit. The D.C. component readily flows through coil L to the power supply, but the fundamental A.C. component encounters a high resonant resistance; part of this A.C. component is absorbed by the tank circuit, and the remainder by the tube.

[•] The efficiency of an R.F. amplifier is the A.C. power output (in the tank circuit) divided by the D.C. power supplied to the plate circuit; multiply this ratio by 100 to get per cent efficiency. The power supplied to the plate circuit is, of course, the D.C. plate voltage multiplied by the D.C. plate current.

The harmonic components find condenser C to be a low reactance path; since only the tube is offering impedance to the harmonic currents. the entire harmonic power is wasted in the tube. A tank circuit having a high Q factor will have an unusually high fundamental A.C. component flowing through both L and C.





FIG. 25. Circuit diagram and characteristic curve for a typical triode tube operating as a class B amplifier.

FIG. 26. Circuit diagram and characteristic curve for a typical pentode tube operating as a class C amplifier. The negative half-cycles at x have been omitted to conserve space on the drawing.

Even greater operating efficiencies (as high as 85%) can be obtained by making the C bias more negative than the cut-off value (usually about twice the cut-off value, as in Fig. 26) and increasing the grid excitation until the maximum possible plate current is obtained; the tube is here operating as a class C amplifier. Notice that the resulting plate current pulses in Fig. 26 are almost rectangular in shape, and flow for considerably less than a half cycle. Experiments have shown that



FIG. 27. Typical push-push power R.F. scoplifier circuit.

with this wave form, more of the desired A.C. power is absorbed by the tank circuit.

Class C operation is widely used in power R.F. amplifiers where high efficiency and a large amount of power are desired. With both class B and class C operation, coupling the load to the tank circuit coil insures that the load will be fed with power at the fundamental frequency.

Self-Biased R.F. Amplifier Circuits. Although a separate C bias supply (like that in the circuit of Fig. 25) is generally used in power R.F. amplifiers, a self-

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biased circuit like that in Fig. 26 is occasionally used. Here a pulsating rectified D.C. grid current flows through grid resistor R when the grid is driven positive by the excitation, thus making that resistor terminal negative which is closer to the grid. Condenser C in Fig. 26 provides a low reactance path for the A.C. excitation being applied to the grid and at the same time smooths out the pulsating direct current flowing through R, making the C bias voltage constant in value.

Power Amplification. Since the grid in the circuit of Fig. 26 is drawing current, power is being absorbed by the grid-cathode path in the tube; this power must be supplied by the grid excitation source. The power delivered to the plate tank circuit divided by the power fed to the grid gives what is known as the power amplification of a high-power R.F. amplifier stage. The power amplification value is usually about 10, which means that in order to get 1,000 watts out of a power R.F. amplifier, you must furnish about 100 watts of power to its grid.

Push-Push Power R.F. Amplifiers. In audio frequency amplifiers, a circuit containing two tubes arranged for push-pull or push-push operation is widely used to secure large power outputs, as you already know. These same two-tube circuits can, with slight modifications, be used to deliver high power output as R.F. amplifiers. A typical push-pull or push-push R.F. amplifier circuit is shown in Fig. 27. Split-stator condenser C_1 divides the input voltage between the two tubes. A fixed C bias voltage is applied to the grids through an R.F. choke and coil L_1 . Each tube feeds into the resonant load circuit made up of center-tapped coil L_2 and split stator condenser C_2 .

TEST QUESTIONS

Be sure to number your Answer Sheet 16FR-1.

Place your Student Number on every Answer Sheet.

Never hold up one set of lesson answers until you have another ready to send in. Send each lesson in by itself before you start on the next lesson. In this way we will be able to work together much more closely, you'll get more out of your Course, and you will receive the best possible lesson service.

- 1. What four things must the R.F. amplifier do in a sound or television radio receiver?
- 2. What does a frequency response curve (or resonance curve) for a tuned R.F. amplifier show?
- 3. In the modern definition for good selectivity, how many times more should the desired signal frequencies be amplified than the nearest undesired signals?
- 4. How does the cascading of R.F. amplifier stages affect the amplification and selectivity of an R.F. amplifier?
- 5. When a tuned transformer load is used in a single-tube R.F. amplifier stage, under what condition is maximum voltage amplification obtained?
- 6. Of the four circuits shown in Fig. 9 (A, B, C and D), which one uses a series resonant circuit?
- 7. Which inter-electrode capacity in a triode tube causes regeneration or oscillation?
- 8. In an R.F. amplifier like that shown in Fig. 14A, what happens when the screen grid by-pass condenser C_2 opens?
- 9. What does the suppressor grid do in a vacuum tube?
- 10. Will variable-mu (super-control) pentode tubes reduce cross-modulation and modulation distortion?

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FEAR LEADS TO FAILURE!

No matter how hard a person may work for success, there is nothing which can help him if he is always doubting his own ability — if he is always thinking about failure.

To be ambitious for wealth yet always expecting to be poor is like trying to get past a vicious dog when afraid of the dog and uncertain of your ability to make friends with him—in each case, fear of failure is almost certain to result in failure. Success, on the other hand, is won most often by those who believe in winning.

Never doubt for a moment that you are going to succeed. Look forward to that success with just as much assurance as you look forward to the dawn of another day, then work—with all that's in you for success.

J. E. SMITH



STUDY SCHEDULE NO. 17

For each study step, read the assigned pages first at your usual speed. Reread slowly one or more times. Finish with one quick reading to fix the important facts firmly in your mind, then answer the Lesson Questions for that step. Study each other step in this same way.

I. The Fundamentals of Amplitude Demodulation.....Pages 1-8 After reviewing some of the fundamentals of amplitude demodulation, you learn that detection or demodulation consists of two steps: rectification and signal separation. Then you study the two basic methods of signal separation and learn some of the requirements that must be met by practical detectors. Answer Lesson Questions 1 and 2.

- - 8. Reviewing Methods of Providing C Bias Pages 31-36 Now that you have studied most of the basic radio stages, this review of battery, self, and fixed bias will help you understand some of the many circuit variations you will meet. Answer Lesson Question 10.
 - 9. Mail Your Answers for this Lesson to N.R.I. for Grading.
- 10. Start Studying the Next Lesson.

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HOW DETECTORS WORK IN RADIO AND TELEVISION RECEIVERS

The Fundamentals of Amplitude Demodulation

THE radio signals that are picked up by the antenna of a sound or television radio receiver are modulated radio frequency signals, which are produced at the transmitter by varying an r.f. carrier in accordance with a sound or picture intelligence signal. This results in an r.f. signal that has either an amplitude or a frequency variation corresponding to the intelligence signal. Notice that the intelligence signal, as such, no longer exists. It has been converted into a variation in the r.f. carrier, and this variation must be reconverted at the receiver into a duplicate of the original intelligence signal. This reconversion is called demodulation or detection, and is performed by a demodulator or detector stage in the receiver. Essentially, a modulated r.f. signal is fed into the demodulator, and a copy of the intelligence (audio or video) signal is obtained from it. (Superheterodyne receivers use an extra detector which serves another purpose; it is called a beat frequency or heterodyne detector. We will study first the demodulator, then this heterodyne detector.)

► As there are two systems of modulation (a.m. and f.m.) in wide use today, each requiring its own type of demodulator, we must divide detector circuits accordingly. In this lesson, we shall first describe the basic facts underlying all detectors. Then, most of the remainder will be on detectors intended for amplitude - modulated sound waves, although we shall briefly cover several basic f.m. demodulators as well as a basic television detector. (Later in the Course, there will be complete lessons on f.m. and television receivers; in those lessons, these demodulators will be covered in greater detail.)

► To understand more clearly just what is necessary for detection, let us review a few facts about the process of amplitude modulation before we take up a detailed study of how detectors work.

MODULATION PERCENTAGE

When the broadcast station to which a receiver is tuned becomes silent for a few seconds—say just after the station announcement—the r.f. *carrier* voltage that reaches the detector input will be unmodulated. The wave form of this unmodulated carrier is shown in Fig. 1A. The peak value N in this drawing is constant and is a measure of the *intensity* level of the carrier.

If a pure sine-wave sound is picked up by the microphone after the silent period, the carrier will be modulated with this sine-wave signal and, in the process of amplitude modulation, it will have the wave form in Fig. 1B. M represents the peak value of the sinewave modulation signal; the ratio of M to N is a measure of the amount of modulation. In fact, M divided by Nand multiplied by 100 gives the percentage of modulation. For example, if the peak carrier voltage level N in Fig. 1B is 10 volts and the peak modulation voltage M is 4 volts, the percentage of modulation will be $(4 \div$ 10) \times 100, or 40%.

A line drawn through the peaks of the r.f. carrier cycles (such as the dotted lines in Figs. 1B, 1C, and 1D) is called a *modulation envelope*. This envelope, which represents the manner in which the r.f. signal is being level, the greater will be the demodulated output of the detector in a receiver. However, modulation percentages greater than 100% are prevented at the transmitter, because greater percentages cause interference with other channels and produce distorted signals.

There are many different modulation envelope shapes since they depend on the modulation voltage and its variations. One possible envelope is shown in Fig. 2A. If a signal with this wave form is fed into an r.f. amplifier stage having a stage gain of 3, the signal that enters the detector will be like that in Fig. 2B, with three



FIG. I. The unmodulated r.f. carrier at A takes on the wave forms at B, C, and D when modulated with a sine-wave signal at various modulation percentages. These waves can represent currents as well as voltages.

varied, is an *imaginary* line representing a variation of exactly the same frequency as the modulating sound. It therefore *represents* the intelligence signal that we want to obtain from the carrier.

When the carrier level at the transmitter is held constant and the modulation signal level is increased, the percentage of modulation increases. When the two voltages, M and N, are exactly equal, there is 100% modulation, and the wave form will be like that in Fig. 1D. As we shall see, the greater the percentage of modulation of a transmitter for a given carrier times the amplitude but exactly the same wave form as the input signal at A. Both carrier and modulation are amplified equally, so the percentage modulation remains unchanged after amplification.

In the study of detectors, just as in the study of other radio circuits, it is much simpler to consider only sinewave modulation signals and sinewave carrier signals. If the detector will handle a sine-wave modulated signal without distortion, then it will handle the corresponding component of a complex wave equally as well.

Demodulation Steps. Now, let's

see just what a detector is required to do. We know that the intelligence signal exists only as an r.f. variation. Our detector circuit, then, must respond to the *variations* in the carrier. In other words, when we feed a varying r.f. signal into our detector, the output voltage of the detector must be proportional to the variations in the signal (that is, this voltage must follow the modulation envelope). The output of the detector will then be the intelligence signal that we want.

As you see from Fig. 1, there are two modulation envelopes in the r.f. signal, one drawn through the positive r.f. peaks and the other drawn through the negative r.f. peaks. One is the exact opposite of the other and, as





shown in Fig. 2, an amplifier stage amplifies both of them. However, the demodulator must follow the variations of only one of the modulation envelopes, so we must get rid of one of them. We do so by *rectifying* the r.f. signal. By cutting off half of each r.f. cycle, we will obtain a series of pulsations having peak values that follow one modulation envelope. Then, we use a *filter* to obtain the intelligence signal from these pulses.

► To sum up: To get exactly the signal we want from an amplitude





modulated r.f. carrier, we must do two distinct things to the r.f. signal. First, the incoming signal is rectified, so that half of each cycle is removed. The result is a series of r.f. current pulses that vary in accordance with the wave form of the intelligence signal. This rectified signal is then passed through a filter which, in effect, separates the desired intelligence from the undesired r.f. current. Keep these two steps (1, rectification; and 2, signal separation) in mind as you study the various types of detector circuits.

(Incidentally, some engineers call signal separation "restoration," since the intelligence signal is re-created from the r.f. pulses by the process.)

RECTIFICATION

Suppose we see what happens when we feed r.f. signals into a simple halfwave rectifier circuit like that shown in Fig. 3. This circuit consists of a diode rectifier, a load resistance $R_{\rm L}$, and a transformer through which we introduce the r.f. We will assume that the diode has a linear (perfectly straight) characteristic curve.

As you know from your previous



FIG. 4. The results of rectification of various signals are shown at B, D, and F.

study of half-wave rectifiers, if an unmodulated r.f. carrier (Fig. 4A) is fed in, the plate current will consist of a series of pulses (Fig. 4B).

Now, suppose we feed in a signal like that shown in Fig. 4C. The peak value of this signal is varied suddenly; first it is like that in Fig. 4A, then it is tripled in magnitude, and finally it is made double the peak value of the original. The rectified plate current (Fig. 4D) will again consist of pulses, this time of three different sizes that correspond to the three peak values of the unrectified signal voltage.

Now, let us feed an r.f. carrier modulated by an audio-frequency sine-wave voltage (Fig. 4E) into our rectifier. This modulated carrier is much like the carrier in Fig. 4C, except that its peak values vary more smoothly, following the shape of the modulating sine wave, instead of increasing or decreasing in sudden steps. When we rectify this sine-wave modulated signal, we get a series of pulses in the plate current (Fig. 4F), varying in peak_value as the peak value of the modulated r.f. signal varies.

Thus, passing an r.f. voltage through a rectifier will give a series of halfwave r.f. pulses with an envelope or outline that follows the shape of the envelope of the modulated signal.

This series of pulses is still not the intelligence signal that we want, but an analysis of these pulses would show that we have a complex wave containing the desired signal. As shown in Fig. 5, the pulses at A have at least three components: (1) a steady value of d.c. (Fig. 5B); (2) an a.c. component that has the same frequency as the modulating sine wave (Fig. 5C); and (3) an a.c. component that has the r.f. carrier



FIG. 5. A rectified wave has d.c., a.f., and r.f. components, as shown here-

(Fig. 5D). (In addition, there will be a number of harmonics of the r.f. carrier that we can ignore.) Therefore, all we need now is some way to separate the component we want from all the others.

► The currents and voltages in an actual detector circuit are never this simple, principally because, as you will learn a little later, we seldom get perfectly linear rectification. Non-linear rectification tends to distort



FIG. 6. This is a complete diode demodulator of a basic type. This circuit can rectify and separate the signal components so that the desired intelligence signal is available across $R_{\rm L}$.

the wave form of the intelligence signal, adding harmonics to it. However, here at the beginning of your study of detectors, we will assume that we have perfect conditions and will leave the distortions produced in practical circuits for discussion in later sections of the lesson.

SIGNAL SEPARATION

The rectifier circuit shown in Fig. 3 has accomplished one of the steps of demodulation. We must now separate the components of the pulses so as to obtain the desired intelligence signal. This is easily accomplished by using a filter which, in its simplest form, consists of a condenser placed across the load resistor as shown in Fig. 6.

This gives us the equivalent circuit shown in Fig. 7. The pulses can be considered as coming from the source E, and as being fed through the plate resistance r_p to the combination of R_L and C. Since the operation of this circuit depends to some extent on whether the value of r_p is high or low, we'll study its workings under both conditions. First, let's see what happens when r_p is high.

The R-C Filter Action. If the value of r_p is high, and C is chosen to have a low reactance at the carrier (r.f.) frequency, then r_p and C will form an ordinary resistance-capacitance filter. Most of the r.f. component will be dropped across r_p and little will be left across C and $R_{\rm L}$.

If C is properly chosen, it will have little effect on the intelligence frequencies, and none on the d.c. (assuming that there is no leakage), so these two components will appear across $R_{\rm L}$. Of course, $R_{\rm L}$ and $r_{\rm p}$ form a voltage divider for these components, but if $R_{\rm L}$ is much higher in value than $r_{\rm p}$, then most of these voltages will appear across $R_{\rm L}$. (Since the condenser offers a lower impedance to the r.f. carrier than it does to audio frequencies, we may say that the condenser



FIG. 7. The equivalent circuit for the signal . separation function of a diode demodulator.

"by-passes" the r.f. pulsations around resistor $R_{\rm L}$, leaving the d.c. and audio components.)

The value chosen for C is important. The larger the capacity the better the r.f. components are filtered out, but C then filters out the higher intelligence frequencies as well. It is necessary to have C small enough so that the desired intelligence signal components will not be lost, even though this means less perfect r.f. filtering.

► When the filtering is accomplished as just described, by an R-C filter, the current through $R_{\rm L}$ is the average of the diode pulsations, as shown by the heavy line in Fig. 8. The faithfulness with which this average value follows the peak value of the pulses determines the amount of distortion introduced by the detector, as we shall learn later.

The Condenser-Input Filter. When the value of r_p is very low, as it is in most diode tubes, we have to consider the action in Fig. 7 in a different manner. We now have to consider the tube as an electronic switch that completes the circuit through its low resistance whenever its plate is positive with respect to its cathode, but opens the circuit whenever the polarity is reversed.

Now, the condenser acts very much like the input condenser of a powerpack filter. It is charged up rapidly through the relatively low resistance of the tube when tube current flows, but, when the current is cut off, Cmust discharge through resistor $R_{\rm L}$. The rate of discharge depends on the



FIG. 8. How the voltage across R_L follows the average of the rectified pulses when the plate resistance of the circuit in Fig. 7 is high.

time constant of $R_{\rm L}$ and C and (since $R_{\rm L}$ is a high resistance) it is far slower than the charging rate. This rapid charge — slower discharge — action means that the condenser voltage tends to follow the peaks of the r.f. pulses, as shown in Fig. 9. Hence, it almost exactly follows the peak value of the modulation envelope. Thus, it "smooths out" the r.f. pulses, leaving only the intelligence and d.c. components.

As the condenser voltage is also the

voltage across $R_{\rm L}$, the load voltage also follows the intelligence signal instead of the r.f. signal—in other words, we have signal separation.

▶ It is usually better to have r_p low in value, rather than high. If r_p is low in value, the load voltage follows the peaks of the pulsations, and we ob-



FIG. 9. A low value of plate resistance in the circuit in Fig. 7 permits the R_L voltage to follow the peaks of the rectified pulses.

tain a higher voltage across $R_{\rm L}$. Further, as you will see a little later, we can get less distortion when $r_{\rm p}$ is low.

THE DETECTOR OUTPUT

After signal separation — whether we follow the average or the peaks, or have a combination of the two actions —we will have across R_L an intelligence signal voltage, a d.c. voltage, and perhaps a small r.f. ripple. The signal across R_L is then fed to a low frequency amplifier—usually through a coupling condenser that blocks the d.c. component, passing only the desired intelligence signal (and a small r.f. ripple which we can ignore). As we shall see later, the d.c. component of the signal across the detector load is useful for control purposes.

► Incidentally, a certain amount of signal separation can occur in the lowfrequency amplifier. However, if we apply high r.f. voltages to the lowfrequency amplifier it may drive the first audio tube off the straight portion of its characteristic. This will introduce distortion, so we wipe out as much r.f. as possible before applying the signal to the low-frequency amplifier.

► This completes the story of the

basic steps of demodulation. However, there are many characteristics of detectors that we have not yet touched upon---distortion, for example. Suppose we now find out about some of the requirements placed on detectors, then go on to a study of typical circuits.

DETECTOR REQUIREMENTS

There are three important performance characteristics on which detectors are rated: <u>sensitivity</u>, fidelity, and voltage-handling ability (sometimes called power-handling ability).

Sensitivity. Radio men are always interested in how much of the intelligence signal voltage they can obtain from a detector when it is fed an r.f. voltage at a definite carrier level and a given percentage of modulation. The sensitivity rating of a detector gives this information, for it is the ratio of the intelligence voltage output to the modulated r.f. voltage input. The more sensitive a detector is, the higher will be its output voltage.

With modern high-gain circuits, we are able to get more than enough amplification, so that we can afford to use detectors having no gain. However, in the early days of radio, before high-gain tetrode and pentode tubes were developed, high sensitivity was one of the most important requirements of a detector.

► We might take the time to cover the history of detectors briefly and see how this requirement has changed. The diode tube was used as a detector as soon as it was developed, but it was dropped after the invention of the triode, because triode detectors gave both amplification and detection. The triode was used first in a form of the C-bias circuit, but this was replaced soon by the grid leak-condenser square-law detector circuit which gave a high output (along with severe distortion). When even more gain was sought, regeneration was added to this circuit. (These circuits will be studied later in this lesson.)

With the development of stable r.f. circuits and more powerful transmitters, there was less need for high gain, and the public demanded better fidelity. First regeneration was dropped, then the linear C-bias detector was developed. This type provided gain and improved fidelity.

Today, the diode has replaced both the grid leak-condenser and the Cbias types, as there is no longer any need for gain in the detector stage. These older types are of interest to us only because receivers using them are still being serviced. (The early crystal detectors are not found in broadcast receivers except in the simplest types intended for school-boys. However, a modified type of crystal detector is now coming into use as an ultra-high frequency detector.)

Fidelity. A transmitter is considered to have high fidelity when the envelope of the modulated-carrier current is a true reproduction of the intelligence signal. The receiver r.f stages can alter the wave, particularly by side-band cutting, so it is only fair to consider detector fidelity in terms of the signal fed into it, rather than in terms of the radiated wave. Hence, we compare the wave form of the output of the detector with that of the envelope of the modulated signal fed into it; the more nearly these two wave forms correspond, the better is the *fidelity* of the detector. High fidelity—in other words, lack of distortion -is obviously desirable in a detector.

Voltage Operating Range. The range of input signal voltages over which a detector will operate with satisfactory fidelity is by no means unlimited. Some detectors can handle only weak carrier signals without distortion, while others will distort on weak carriers yet work perfectly well on strong carriers. The percentage of modulation of the incoming signal also limits the operating range of a detector; some circuits work well at high percentages of modulation, while others do not.

► Let us now study some of the practical detector circuits, and learn how they compare in their performance characteristics.

How Diode Detectors Work

We used a diode detector in the preceding section to explain the fundamentals of detection. However, in that preview, we made a number of assumptions and told only the basic story. We'll now study the details of diode detector operation.

R.F. LOADING

Fig. 10 shows a diode detector circuit that is typical of those in modern superheterodyne receivers. The i.f. transformer feeding the detector is almost always tuned, so we can represent it by L_1 - C_1 and L_2 - C_2 . The intelligence signal developed across the load R_L is fed through coupling condenser C_4 to the low-frequency amplifier.

The resonant circuit L_2 - C_2 acts as the source, and must supply power because the diode tube passes current whenever its plate is positive with respect to its cathode. This "loading" reduces the tuned circuit Q because the effect is that of connecting a resistance across the L_2 - C_2 terminals. As a reduction in Q reduces both the output and selectivity, we cannot permit too much loading.

As shown in Fig. 11, the diode plate resistance r_p and the diode internal plate-to-cathode capacity C_{PK} form a parallel group which is in series with the R_L-C_3 group, with the entire combination connected across the terminals of L_2 - C_2 .

We can assume that, at the carrier frequency, the reactance of C_3 is very low so that R_L is effectively shorted for r.f. This leaves r_p and C_{PK} across the resonant circuit. We can dismiss the effect of C_{PK} because it serves only to detune the resonant circuit and we can adjust C_2 to compensate for it. Hence, the actual loading depends on the value of r_p .

Fortunately, the effective value of $r_{\rm p}$ is increased and the loading is reduced by the action of C_3 and $R_{\rm L}$. You know that both the intelligence frequency component and the d.c. component are stored in C_3 , with the polarity shown in Fig. 10. Hence, we have two voltages applied to the diode stored in C_3 . Since the C_3 voltage is only slightly less than the peak value of the incoming signal, and since its polarity tends to make the diode plate negative, the tube can conduct only during the relatively short time that the incoming signal is able to overcome the C_3 voltage and make the plate positive. The plate current becomes a series of short, high pulses having a low average value.

Increasing the values of C_3 or R_L will increase the time constant and will cause the C_8 voltage to follow the peak value of the incoming signal even closer. This reduces the plate current and the loading even further. However, as we will show, this increase in time constant lessens the ability of the circuit to pass the higher intelligence frequencies; this sets a limit to the values we can use. For this reason, we must accept some loading when we use a diode detector. This is not too serious since modern i.f. amplifiers usually provide enough selectivity and gain to make up for most of the effects of loading.

Service Hint. Notice the importance of condenser C_8 to the r.f. circuit. If C_3 opens, several things happen. The shunting effect of the internal tube capacity is reduced and, in addition, the loading on the tuned circuit is changed. More important is the fact that there is now a voltage divider (the tube and R_L) for r.f., and a large portion of the signal voltage will be dropped across R_L instead of being applied to the tube. As a result of this reduced r.f. voltage across the



FIG. 10. The input and output connections of a diode demodulator.

tube, the plate current pulses will be far smaller, so the output will be reduced greatly. And, of course, the absence of C_3 means that the detector output is now a series of r.f. pulses, and signal separation can occur only in the low-frequency amplifier.

► When you find that the output from a detector is far below normal, you should first suspect a defective

diode tube, then check the alignment. Keep C_8 in mind, however, as it may be defective.

FIDELITY CONSIDERATIONS

Naturally, we are always interested in how faithfully a detector reproduces the original audio signal — or, to put it another way, how much distortion is present in the output of a detector. There are two possible causes



FIG. 11. How the diode circuit can act as a load on the input resonant circuit.

of distortion in the output of the diode detector we've just studied: (1) there will be distortion if the diode is not perfectly linear in its operation, and (2) there will be distortion if all the audio signal is not reproduced across the load. The first of these is primarily caused by the tube, while the second is primarily a function of the load and r.f. filter.

NON-LINEAR TUBE CHARACTERISTIC

It's easy to see why a tube that is not linear in its response will cause distortion. A diode tube has linear response if its plate voltage-plate current characteristic is a straight line, because then equal changes in plate voltage will produce equal changes in plate current. The response of such a tube to a modulated r.f. carrier is shown in Fig. 12. As you can see from this figure, the envelope of the rectified pulses is exactly like the envelope of one half of the input signal. This means the a.f. current produced by the tube has the same wave shape as the
modulation envelope, and, if the voltage across the tube load has the same form as the tube current, there will be no distortion caused by the tube.

If the tube characteristic is not linear, but is instead curved like that in Fig. 13, then the envelope of the rectified pulses is no longer precisely like the carrier envelope; therefore, the a.f. voltage developed across the load resistor will not have the same form as the original audio. Distortion will have been introduced by the nonlinear response of the tube. This distortion consists principally of the addition of *even* (the second, fourth, etc.) harmonics to the original signal.

Modern diodes have relatively straight characteristics, and the use of high load resistor values tends to straighten them out further. Even so, there is usually some curvature near the zero-current point of the curve. Let's see what this can cause.

MODULATION PERCENTAGE AND SIGNAL INPUT LEVEL

The amount of signal and the percentage of modulation have much to do with how much distortion may occur in diode detectors.

As we have shown, most diodes have characteristics that are relatively straight except at the bottom, where



FIG. 12. The output follows the input exactly when the detector characteristic is perfectly linear.

they curve like the characteristic in Fig. 13. Such a tube will give undistorted response as long as the modulation envelope (the peaks of the r.f.) swings over the linear part of the characteristic, but_will have_a somewhat distorted response if the modulation envelope swings down over the curved part of the characteristic.

Now, the percentage of modulation



FIG. 13. A curved tube characteristic causes distortion.

partly determines whether or not the envelope will reach the curved part of the characteristic. Assuming we have only 40% modulation, like the wave in Fig. 1B, and the signal is fairly strong, the envelope does not reach the curved section at the bottom of the characteristic, and the detector output therefore will be undistorted.

If, however, the carrier is 100% modulated, as in Fig. 1D, the modulation envelope cannot help but swing down over the curved part of the characteristic. Part of the wave will then be distorted.

► The strength of the signal determines how bad the distortion will be. If the signal is weak, even the highest signal peaks will not go very far up on the characteristic. Consequently, the modulation envelope will swing mostly over the curved part of the characteristic. There may be severe distortion, even if the carrier is only 40% or 50% modulated. On the other

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hand, if the signal is very strong, practically all the modulation envelope will swing over the linear part of the characteristic. Even if the modulation is 100%, only a very small part of the output will be distorted. Thus, a diode detector produces the least distortion when it is fed a strong signal.

This is a very important fact. Broadcast stations try to use a high percentage of modulation (up to 100%), which, as you just learned, will cause considerable distortion in a diode detector unless the signal is strong. Obviously, then, it is desirable to build up the signal as much as possible before applying it to the diode. This is one of the reasons that the diode detector was not widely used until high-gain tetrode and pentode tubes were developed for use in r.f. and i.f. stages preceding the detector.

Power Detectors. Today, the diode is commonly used as a power detector. This is simply a detector that can handle very large r.f. voltages (from about 5 to 50 volts) without appreciable distortion.

THE R-C TIME CONSTANT

Distortion will be introduced also if the voltage developed across the load resistor does not have the same form as the envelope of the rectified current pulses. This distortion is caused primarily by the time constant of the r.f. filter condenser and the load resistance.

To see what will happen, let's assume that the time constant is too long to follow the modulation frequency shown by the dotted line in Fig. 14A. As you will notice, the condenser voltage follows the peaks satisfactorily between points X and Y of this figure, where either the carrier is not changing or the modulation is swinging positive. When the modulation signal swings in the other direction, however, the condenser voltage decreases along the heavy line between points Y and Z instead of following the dotted line. This means that we get the output wave shown by the heavy line in Fig. 14B instead of the sine-wave signal shown by the dotted line. Higher modulating frequencies than that shown are followed even less well: they are not only distorted, but are also reduced in amplitude. Thus, too high a time constant causes two kinds of distortion: some frequencies are not reproduced as well as others (this is called frequency distortion); and the shape of the output



FIG. 14. If the R-C time constant is too high. a distortion known as "clipping" occurs, as illustrated here.

wave is changed, showing that harmonics have been introduced (amplitude distortion).

We are therefore definitely limited as to the time constant we can use. The values of $R_{\rm L}$ and C_3 must be made sufficiently small to allow the condenser to discharge fast enough to follow the highest frequency that we want.

▶ In practice, design engineers usually choose the value of C first, using a value about 5 to 10 times the tube capacity $C_{\rm PK}$ to minimize the r.f. loss. Then, with this value fixed, they choose the highest value of $R_{\rm L}$ that will give the desired fidelity.

Fig. 15 gives examples of the attenuation introduced by different values of R_1 , when the capacity is fixed at 250 $\mu\mu$ fd. You can see that smaller values of R_L give less frequency distortion. (However, small values of R_L permit more loading on the resonant circuit and more amplitude distortion in the tube.)

► To have high-fidelity detection, we must reduce to negligible values both distortion produced by the tube and distortion produced by the load. Unfortunately, the measures we can take to correct the first tend to accentuate the second. That is, a diode is more nearly linear in its response when the load used with it has a high ohmic



FIG. 15. How increasing the load resistance causes the high frequency response to fall off (a form of frequency distortion).

value, but this means that the time constant (R times C) of our load circuit will be rather high; as you just learned, this tends to cut off the higher audio frequencies.

R.F. Filters. When the r.f. by-pass condenser (C_3 in Fig. 10) is the only r.f. filter, its capacity must be kept high so that a minimum of r.f. voltage will be applied to the low-frequency amplifier. One way to keep this r.f. voltage low, and at the same time use a smaller capacity, is to use a complete r.f. filter as shown in Fig. 16. Here, C_2 acts as an input filter condenser, while the R_1 - C_3 combination forms a standard R-C filter section. (An r.f. choke coil can be used in place of R_1 , but it is less expensive to use the resistor and the results are just as good.)

The improved filtering given by the R-C filter makes it possible for condensers C_2 and C_3 to be smaller. In fact, their sum can be less than the value needed for C_3 in Fig. 10. This means that the higher intelligence frequencies are not cut off as much. This scheme allows higher fidelity, and permits the use of load values up to 500,000 ohms.

► It is possible to reduce the distortion in a diode detector stage to a very low value by selecting a tube that has a nearly linear characteristic and by choosing the values of the load with care. (The human ear will usually put up with more frequency distortion than amplitude distortion, so usually the load resistance is kept fairly high to reduce the amplitude distortion in the tube.) As a practical matter, this means that a diode detector can have virtually perfect fidelity, because even a highly trained ear cannot distinguish small amounts of distortion.

ISOLATION OF VOLUME CONTROLS

Now, let's take up a variation in the connections that you will encounter one that arises because many receivers use volume controls as the diode detector load. The control is used to vary the input voltage fed to the low-frequency amplifier.

A volume control has a movable slider which will eventually wear out the resistance element. The resulting poor contact will produce noise. These controls become noisy much more quickly when d.c. current flows through them. As you know, a d.c. current flows through the diode detector load resistance because the filter circuit removes only the r.f. pulses and leaves the low-frequency and d.c. components of the signal. To increase the life of the volume control, some manufacturers isolate the control from this d.c. current.

Fig. 17 shows the usual connection with the volume control R_c used as a diode load. The coupling condenser C_2 feeds the low-frequency component of the signal to the lowirequency amplifier, blocking out the d.c. from the amplifier input but not from the volume control.

By adding another blocking condenser and another resistor, as shown in Fig. 18, it is possible to keep d.c.



FIG. 16. The C₂-R₁-C₈ filter shown here is found in most receivers, since it reduces the r.f. component without so much loss of the higher intelligence frequencies.

out of the volume control.^{*} Notice that the added resistor R_1 is used as the diode load resistor, and the additional coupling condenser C_3 passes only the low-frequency signal to the volume control R_c . From here, condenser C_2 passes on the low-frequency signal to the low-frequency amplifier.

With this arrangement, only the low-frequency signal flows through R_c . The d.c. in the diode circuit flows



FIG. 17. Volume controls are frequently used as the detector load, connected in the manner shown here.

through resistor R_1 , not through the volume control.

When the volume control is used as the detector load as in Fig. 17, its value is chosen just as that of a fixed load resistor is.

► When the control is isolated as in Fig. 18, however, it is desirable to use a higher resistance control than that used in Fig. 17. As you know, the reactances of C_3 and C_2 are negligible at the middle audio frequencies. This means that, as far as a.c. is concerned, $R_{\rm c}$ (and any resistance between the output terminals) will be effectively in parallel with R_1 . Since two resistors in parallel have a total resistance that is lower than either of them has individually, there must be an increase in the values of R_c and R_1 to make their total resistance higher, or we must accept a lower a.c. load resistance in the diode circuit.

A lower a.c. load would mean that



FIG. 18. There will be far less noise and longer control life when the volume control is isolated so that d.c. current does not flow through it.

^{*}Of course, the circuit shown in Fig. 17 costs less than the set-up in Fig. 18. That is why the former is so widely used.

less of the signal would be developed across the load, and also that a certain amount of amplitude distortion would occur.

► As a practical case, the volume control resistance in Fig. 17 is commonly 250,000 ohms. However, if the volume control is isolated for longer life by using the circuit in Fig. 18, then resistor R_1 may be about 250,000 ohms and the volume control R_c may be between 500,000 ohms and 1 megohm. (The additional coupling condenser C_3 is usually from .05 mfd. to .1 mfd.)

Practical Diode Circuits

In most practical diode detectors, the d.c. voltage developed across the load resistor is used for control purposes. A typical circuit is shown in Fig. 19.

In this figure, tube VT_2 is the diode detector. Its circuit and action are precisely like the diode we just described.

The amplifier stage using VT₁ is also standard, except that its grid return lead has R_2 in it and is connected to the diode load resistor R_1 instead of going directly to ground. Tracing from the amplifier cathode (point 1), you see that the bias voltage developed across R_s is applied to the grid through R_1 and R_2 . (Trace from point \mathscr{L} through the chassis to point \mathscr{J} , then to points 4 and 5.)

The amplifier stage uses a variablemu or super-control tube, so the amplifier gain will vary with the bias. The gain is maximum when the bias is minimum. In Fig. 19, the gain will be maximum when no signal is tuned in, for then the only bias is the self-bias developed across R_{e} .

However, when a signal is received, a d.c. voltage will appear across R_1 with the polarity shown in Fig. 19. (Check the polarity by tracing the direction of electron flow in the diode circuit.) This voltage, whose value depends upon the strength of the carrier voltage applied to the diode, adds to the R_s bias voltage (trace from point 1 to point 4: the two voltages are in series, connected so that their voltages add). This increase in bias reduces the gain of the amplifier.

It is possible to choose part values for these circuits that will keep the



FIG. 19. A basic a.v.c. circuit.

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level of the signal applied to the diode at a nearly constant value. If a strong signal is received, the d.c. voltage drop across R_1 will be high; this will increase the bias on the amplifier tube and reduce its gain sufficiently to bring the signal applied to the diode down to the desired level. Conversely, the d.c. voltage across R_1 will be low when a weak signal is received; this



FIG. 20. How a multi-section tube can be used as: demodulator; a.v.c. diode; and first audio amplifier.

will decrease the bias on the amplifier tube and increase its gain, and thus bring the signal applied to the diode up to the desired level. This action will keep the output volume of the set practically constant on local and near-distant stations. That is why it is given the name of "automatic volume control" (abbreviated a.v.c.).

Resistor R_2 and condenser C_4 serve a very important purpose in connection with a.v.c. action. Remember that the voltage across R_1 is made up of two components, a d.c. voltage whose value varies with the carrier level, and a low-frequency a.c. voltage corresponding to the intelligence signal. Clearly we cannot allow the a.c. component to bias the grid of the amplifier, for it would modulate the amplifier input. R_2 and C_4 form an R-C filter section that keeps the a.c. voltage developed across R_1 from reaching the amplifier. (This filter has no effect upon the d.c. component of the voltage across R_1 , for no direct current is drawn through R_2 by the grids of a.v.c.-controlled tubes.)

► A later lesson will treat a.v.c. systems in detail. Here we have presented the subject briefly to give you some idea of what a.v.c. is and how it is secured from a diode detector. The fact that we can get this control voltage from a diode is one of the reasons for the great popularity of this tube.

MULTI-FUNCTION TUBES

Tubes with twin diode sections and a triode or pentode section are often used in radio receivers to secure de-



FIG. 21. Another multi-section tube connection in which the diodes are connected in parallel and a.v.c. voltage is obtained from the demodulator circuit.

tection, a.v.c. voltage, and amplification from a single tube. There are many ways of using a three-in-one tube such as this; Figs. 20 and 21 show two of the more common methods.

▶ In many of the early a.v.c. circuits the required bias voltages were applied through the a.v.c. network, making it desirable to use a separate diode for production of the a.v.c. voltage. Fig. 20 shows a circuit in which diode D_1 is the detector, while diode D_2 is used as a rectifier for the production of the a.v.c. voltage.

When a signal flows through L_1 , a corresponding signal voltage is induced in L_2 , which is tuned by condenser C_1 . This applies an r.f. voltage directly to diode D_1 . The pulses are fed through the r.f. filter C_2 - R_1 - C_3 so that only the d.c. and a.f. components exist across the load resistance R_2 . The lowfrequency component is passed on through C_4 to R_3 , where it is applied to the grid of the triode section of the multi-function tube. The triode then acts as the first audio amplifier.

Condenser C_5 is used to couple some of the r.f. potential into the circuit of diode D_2 . As a result of rectification, a pulsating voltage exists across R_4 , which is filtered of all except its d.c component by the filter composed of resistor R_5 and condenser C_6 . This d.c. component is then applied to the a.v.c.-controlled tubes.

▶ In sets in which no additional biases are applied to the controlled stages, the same type of tube can be used with the extra diode connected in parallel with the detector diode, as shown in Fig. 21. The parallel diodes act as one diode with only half the usual plate resistance. Although this increases the loading on the tuned circuit, it provides a greater output.

The a.f. component developed across R_2 in Fig. 21 is passed through C_4 to the grid of the triode section. The d.c. component, fed through R_4 , gives the a.v.c. action. The filter R_4 -C₅ removes the a.f. and r.f. components.

TELEVISION DEMODULATORS

Fig. 22 shows a typical television detector. This must handle a very

wide frequency band, as the video signal includes frequencies from 10 to 4,000,000 cycles. To get the wide frequency range, very low load resistances—usually from 5000 to 25,000 ohms—are used. Two diodes in parallel are used to keep the output voltage as high as possible (these can be the two elements of a dual-diode tube).

The r.f. filter capacity must be kept small to handle the wide frequency



FIG. 22. A typical television demodulator.

range. As a matter of fact, a condenser may be unnecessary, for sufficient filtering capacity may be provided by the shielded lead going to the grid of the first picture amplifier tube. (Notice that there will be a capacity between the lead and its shield, which is effectively across the load resistance.) ▶ In the circuit shown in Fig. 22, terminal 1 of the load resistance is grounded instead of terminal 2. It is just as common to ground terminal 2 and to take the signal from terminal 1; the particular connection used depends upon the polarity of the signal wanted. (The television signal must be applied to the cathode ray tube with the proper phase, so that increases in brightness will cause brightspots on the cathode-ray tube er screen. As each amplifying stage reverses the phase by 180°, the number of video stages determines which detector connection will give the right polarity.) The subject of television detectors will be covered in another lesson, but you can see now that conventional circuits are used, the major

difference being the use of different part values.

PUSH-PULL DETECTOR

The circuit in Fig. 23A is a pushpull, or full-wave, detector. When point 2 is positive with respect to point 1, the diode D_1 passes current while D_2 is cut off (because point 3 will then be negative with respect to point 1). On the next half-cycle, the polarities



FIG. 23. A full-wave diode demodulator.

of points 2 and 3 are interchanged, and the opposite action occurs. Hence, the diodes alternately pass pulses of current through the load resistor R_1 . This is the same action as is obtained in full-wave rectification, which, as you know, doubles the frequency of the applied a.c. voltage. (A full-wave powerpack rectifier produces 120 cycles from a 60-cycle power line frequency.) In other words, if the r.f. carrier has a frequency of 400 kc., full-wave detection produces r.f. pulses with a frequency of 800 kc.

Doubling the frequency of the r.f. pulses across R_1 does not greatly affect the envelope shape (see Fig. 23B). It does, however, permit a decrease in the size of condenser C_1 , the r.f. filter, because at higher frequencies, condenser reactances are lower. Thus, a smaller condenser can do the filtering.

Since a reduced value for C_1 gives greater fidelity, the push-pull detector has been tried in some high-fidelity receivers. It is not used very often today, however, except for an interesting variation of this circuit that is employed as the demodulator in f.m. receivers.

The C-Bias or Plate Detector

In each of the diode circuits we have discussed so far, none of the tubes is powered by a B supply. The only power fed them (other than filamentheating power) is the modulated carrier voltage. Consequently, our circuits have given us no amplification, for, as you know from your earlier studies, a tube amplifies only if the input voltage is used to control the much larger voltages furnished by a B supply.

This lack of amplification is no problem in modern radios, for we can produce all the gain we need by using amplifier stages before and after the detector. In earlier days, however, when high gain was not so easy to achieve, the diode detector was not in general use; instead, engineers developed circuits that permitted both detection and amplification in the same tube.

LINEAR C-BIAS DETECTOR

The linear C-bias detector is really a vacuum tube amplifier stage supplied with a C bias almost large enough to cut off plate current. It is operated with a plate voltage and a load resistance value which will make the dynamic $E_{\rm g}$ - $I_{\rm p}$ curve of the tube practically straight or linear. (The curve is similar to the linear_type_ shown in Fig. 12.) Automatic C bias, supplied from a resistor in the cathode lead, is almost universally used.

Triode Circuit. An early linear C-bias detector circuit using a triode tube is shown in Fig. 24. Cathode resistor R_c provides an automatic C bias that is high enough to reduce plate current almost to zero when there is no r.f. input signal. When a modulated r.f. signal is applied to L_1 , a voltage of similar wave form is induced in coil L. This voltage receives a resonant stepup from the L-C resonant circuit and is then applied to the grid of the tube.

Since the tube is biased almost to cut-off, plate current flows only when the r.f. input voltage is going through a positive half-cycle. Also, since the dynamic E_{g} - I_{p} characteristic of the tube is approximately linear, these plate current pulses have the same form and envelope as the positive half of the input r.f. voltage. Thus, the output of the triode is like that of the diode we have already studied, except that in this new circuit the pulses have been amplified by the tube. In other words, the tube amplifies in a normal manner, while detection occurs because the grid is biased so near cut-off



FIG. 24. The linear C-bias detector.

that only half of each signal pulse causes plate current to flow.

The primary of the audio transformer T is the load, in place of the resistor used in our previous diode detector examples. Condenser C_1 is the r.f. by-pass condenser used for signal separation. In addition, the r.f. choke



FIG. 25. The equivalent signal-separation circuit of Fig 24.

 L_2 , together with the distributed capacity in the transformer primary, forms a low-pass filter that assists in keeping r.f. from the load.

The equivalent r.f. filter circuit is shown in Fig. 25. We have a filter r_{p} - C_1 and another filter L_2 - C_D . This circuit follows the average of the plate current pulses rather than the peaks, so the load current varies in the manner shown in Fig. 8.

This circuit has the advantage of tube amplification and, as there is nc grid current, it does not load the resonant circuit. However, it does not provide a d.c. voltage for automatic volume control. Also, there are limits to the input signal it can handle; if the signal exceeds the bias, so that the grid is driven positive, the plate current pulses will be distorted.

Pentode Circuit. Another linear C-bias detector circuit is shown in Fig. 26. This uses a pentode tube, and gives relatively high output voltages even for weak input signals. A similar circuit is found today in inexpensive t.r.f. receivers of the a.c.-d.c. type.

While this circuit looks more complex, its operation is much the same as that of the triode circuit in Fig. 24 The control grid is biased practically to cut-off by plate current flowing through bias R_c , which is by-passed by C_1 . R.F. filtering is obtained from C_2 - L_2 - C_3 , which work with r_P to form the filter shown in Fig. 25. The a.f. voltage is developed across resistor $R_{\rm L}$, and is passed on to the next stage through $C_{\mathbf{5}}$. Condenser $C_{\mathbf{4}}$ keeps a.f. out of the power supply and also, in combination with resistor R_1 , filters out any ripple from the power supply. ▶ In each of these C-bias detectors, the modulated signal is rectified because the grid of the tube is biased practically to cut-off-that is, to the point where the $E_{\rm g}$ - $I_{\rm p}$ characteristic makes a sharp bend. For this reason, a C-bias detector is often called a "plate-bend detector" or simply a "plate detector."

SQUARE-LAW DETECTION

All the detector circuits we have discussed so far have two common characteristics. In each, rectification of the modulated r.f. voltage has been



FIG. 26. A pentode C-bias demodulator.

complete (all negative half-cycles have been eliminated) and approximately linear (the envelope of the rectified pulses has been approximately the same shape as the envelope of the modulated carrier).

It is possible to have a detector in which rectification is incomplete and non-linear. Suppose, for example, we



FIG. 27. This drawing shows how the squarelaw detector produces a distorted output signal.

have a triode which has a curved E_{s} - I_{p} characteristic like that shown in Fig. 27. Such a tube is said to have a "square law" characteristic, for its plate current is proportional to the square of the signal voltage applied to the grid. (That is, if the input signal is doubled, the output increases four times; if it is tripled, the output will be increased nine times, etc.)

Suppose, further, that we place this triode in a circuit like that in Fig. 24, and adjust the bias so that the tube will operate at point O in Fig. 27—the point where the E_g - I_p characteristic curves the most. If we now apply a modulated carrier to the input terminals of the stage, we will get the output shown in Fig. 27.

Here, each r.f. cycle has undergone incomplete rectification; each positive swing of the r.f. voltage has been increased, and each negative swing has been decreased, by the curved characteristic. Because of the incomplete rectification, the average value (a.f.) of the rectified current (shown by the heavy line in Fig. 27) does not have exactly the same wave form as the envelope of the positive current peaks.

This means that we will get a somewhat distorted reproduction of the original a.f. voltage when we send the tube output through the signal-separating circuits of the tube load. (The amount of distortion produced depends on the percentage of modulation of the carrier; it is inappreciable formodulations less than 20%, but it goes up to 25% for modulations of 100%.) However, such a detector has the advantage of being very sensitive, for its output current (and therefore its output voltage) is proportional to the square of the carrier voltage. Thus, even very weak signals can be made to give an appreciable output.

► The detector we just described is known as a C-bias square-law detector. The tube used does not have to be a triode; many pentodes also have such a characteristic, and can be used as square-law detectors. In fact, the rectifying device does not even have to be a tube: a galena crystal has a square-law characteristic when used as a rectifier, and, as we shall see later in this lesson, such crystals were widely used as square-law detectors in the early days of radio.

Because they produce distortion, and because modern amplifier circuits make sensitive detectors unnecessary, square-law detectors have practically disappeared as demodulators. However, they are modified into superheterodyne first detectors, and they still are used in measuring instruments and in radio applications where sensitivity rather than freedom from distortion is desired.

Combination Linear-Square Law Detectors. Many tubes, if their operating voltages are properly adjusted, can be made to have an E_{g} - I_{p} characteristic which is square law in shape near the bottom and nearly linear farther up. If we use such a tube in a detector circuit, and adjust the grid bias so that we operate near the point of greatest curvature of the characteristic, we can get a combination of linear and square-law detection. Strong signals will swing over the linear part of the characteristic, and will give practically distortionless detection. Weak signals, however, will swing over only the curved part of the characteristic, giving us square-law detection. We then get sensitive detector action, which is desirable for weak signalsbut we must, of course, pay the price of greater distortion for it.

These combination detectors were once very popular, but, like squarelaw detectors, they have almost entirely disappeared from home radios as demodulators.

Heterodyne or Beat Detectors

The superheterodyne circuit requires two detectors; one is used as a demodulator to obtain the intelligence signal from its carrier (this is the secoud detector), and the other is used to produce the i.f. (this is called the first detector). This first or heterodyne detector is not used to demodulate the incoming signal; instead, it mixes the incoming and local oscillator signals to produce a beat frequency that is still modulated by the intelligence signal.

When two signals of different frequencies are mixed together and then fed into a detector, the resulting rectified current pulses will include the following components: the original frequencies; the sum of these frequencies; their difference; and various harmonics of all these. In the superheterodyne, we are interested in the difference frequency, which is called the intermediate frequency (abbreviated "i.f."). (This difference frequency is also known as a beat frequency or a heterodyne frequency, and the method by which it is produced is called heterodyne detection.)

HOW A HETERODYNE DETECTOR WORKS

The action of a heterodyne detector can be best understood by a study of the wave form curves in Fig. 28. When two r.f. signals that differ in frequency are mixed together, such as the signals in Figs. 28A and 28B, the result is the complex wave shown in Fig. 28C. (Here, the envelope of the wave represents the difference or beat frequency.) Notice that peak a is rounded, while the trough at b is sharp. Clearly, the envelope is not a sine wave.

If a voltage having the wave form shown in Fig. 28C is sent through a linear detector, the resulting demodu-



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lated current shown in Fig. 28D is not sine wave in form. On the other hand, when the wave of Fig. 28C is sent through a square-law detector, the sharp troughs of the demodulated signals are distorted by the detectorcharacteristic, giving the sinusoidal output shown in Fig. 28E. (Here we have a circuit introducing a distortion that compensates for another distortion!) Thus, when the amplitudes of the two signals being mixed are equal, the square-law detector gives a less distorted output than a linear detector.

► However, usually the signals mixed together are not equal in amplitude. Since the amount of heterodyne output depends on the amplitude of the two r.f. signals, the local oscillator output is generally made very strong so that the heterodyne output will be strong even if the incoming signal is weak. Usually the output from the local oscillator is made about 10 times the strength of the strongest expected incoming signal.

-When-the two r.f. signals differ this much in amplitude, the envelope of the mixed signal is practically a pure sine wave. For example, if the local oscillator signal is represented by Fig. 28F and the incoming signal by Fig. 28G, we will obtain the complex wave shown at Fig. 28H when they are mixed. Either a linear or a squarelaw detector will give us a sine-wave output signal from this mixed signal, because it has a sine-wave envelope; therefore, under these conditions, we can use either kind of detector. You will study typical mixer-first detector stages in detail when you study the superheterodyne receiver.

Early Detector Types

We will now take a brief look at some of the early detector circuits. You'll find these no longer used much, but you will meet them occasionally.

CRYSTAL DETECTORS

Before the days of vacuum tubes, practically every radio receiver contained a crystal detector. Even today you can purchase "crystal sets" in many radio stores, for they give acceptable reception of local broadcasts with ample volume for headphone use.



FIG. 29. A typical crystal detector.

Fig. 29 shows a crystal detector circuit that has been widely used. It is tuned by resonant circuits C_1 - L_1 and C_2 - L_2 , which are inductively coupled. R.F. filtering is provided by condenser C and the resistance of the crystal, which form an R-C filter. The crystal is a one-way device; it passes current much better in one direction than in the other. Hence, it provides approximate half-wave rectification of the applied r.f. signal. The audio component of the rectified pulses operates the headphones. Thus, rectification takes place at the crystal, and signal separation occurs at the load (the headphones). There is no amplification in a crystal detector.

The crystals are generally galena, iron pyrites, silicon, carborundum, or combinations of these minerals. A

phosphor-bronze wire, often fine called a "cat's whisker" because it was about as thin and springy as the whisker of a cat, was used originally to make contact with a sensitive spot on the crystal. This was a very delicate arrangement, for the slightest jar necessitated a resetting of the "cat'swhisker." More recently, fixed crystal detectors have been made by mounting a piece of carborundum permanently against a soft metal disc: these do not require adjustment, but are considerably less sensitive (give poorer rectification).

► In certain foreign countries where

was one of the first detector circuits that used a vacuum tube, and was for many years one of the most popular. Two forms of this circuit are shown in Fig. 30; both work in essentially the same manner.

We can consider the cathode and grid of the tube to be a diode in which the grid acts as an anode. Looking at the circuit from this viewpoint, you see at once that the cathode-grid circuit of the tube is really a simple diode detector. R_{g} and C_{g} form the signal separating load circuit, while the resonant circuit $L_{T}-C_{T}$ is the means by which the modulated r.f. carrier



FIG. 30. Two forms of the grid-leak gridcondenser detector.

radio receivers are taxed according to the number of tubes they contain, crystals and special copper-oxide rectifiers (having elements about the size of a pinhead) are still used as detectors.

A recent development is the use of tiny fixed crystal units as detectors at micro-wave frequencies, where the effects of the transit time of electron flow through a tube make tube detectors unsatisfactory.

GRID LEAK-CONDENSER DETECTORS

The grid leak-condenser detector

voltage is introduced into the detector circuit. When a signal is fed in, it undergoes detection in this "diode" circuit, and an a.f. voltage similar in form to the envelope of the carrier is developed across load resistor $R_{\rm g}$. (The resistor $R_{\rm g}$ is called a "grid leak," while $C_{\rm g}$ is known as the "grid condenser;" from these come the name for this detector type.)

The a.f. voltage across R_s is applied between the cathode and grid of the tube, and thus produces a.f. variations in the tube plate current. The r.f. input signal is also applied between the grid and cathode, so the tube plate current contains r.f. variations too. To keep the r.f. voltages out of the low-frequency amplifier, C is used in the plate circuit to form an R-C filter with the tube r_p . The a.f. component is amplified by the tube and appears across the plate load, ready to be fed to the a.f. amplifier.

Thus, a triode grid-leak detector acts like a diode detector to which one stage of audio amplification is added. <u>Rectification</u> and signal separation occur in the grid circuit; the a.f. produced is then amplified by the tube. Proper selection of the resistor R_{g} and condenser C_{g} will let the detector work as a square-law device for tion envelope of the carrier.

One form of regenerative grid leakcondenser detector circuit is shown in Fig. 31. The grid circuit of the tube is exactly the same as that of the grid leak-condenser_detector, but the platecircuit contains a coil L_2 through which the r.f. plate current flows. This induces the desired amount of r.f. voltage in the grid tank circuit, through mutual inductance M. Variable resistor R, connected across the feedback coil, controls the r.f. current flow through the coil and so determines the amount of feedback. (Some receiver builders controlled the amount of feedback by varying the mutual in-



FIG. 31. A regenerative detector.

weak signals or as a linear detector for strong signals.

REGENERATIVE DETECTORS

Even more gain than that found in grid leak-condenser detectors was sought in the early days of radio. Inventors found a way of using the undesired r.f. voltage in the plate circuit —they put this component to work in a regenerative detector, in which the r.f. current in the plate circuit of the detector was fed back into the grid circuit to reinforce the incoming modulated r.f. carrier voltage. Naturally a means of controlling the amount of feedback was necessary, for excessive feedback resulted in oscillation and complete destruction of the modulaductance between coils instead of by varying R.)

The energy fed back from the plate circuit, in effect, reduces the losses in the grid circuit, so a far greater resonance step-up is obtained.

A regenerative detector has some good features, but also has several disadvantages. It is quite selective for the carrier frequency being tuned in, but attenuates side-band frequencies and produces severe frequency distortion. (For code reception and reception of ordinary speech, this frequency distortion is not at all objectionable; hence, regenerative detectors may still be found in certain communications receivers.) Furthermore, a regenerative circuit goes into oscillasily, especially when condjusted from a low to a by. If no r.f. stage is the detector, the resultare radiated by the rejust as if the receiver ransmitter, and intered in neighboring relition, an annoying heard from the rethe frequency of osdiffers only slightly by of the desired ind the resulting beat udio frequency.

> rative Detectors. A ector is "supersensigreatest sensitivity) verge of going into ever, when it is in nall increases in caret the circuit into osthis occurs, decreases isity will not stop the

ty was eliminated with the super-. enerative detector, a circuit based upon the principle that it takes a certain amount of time for a detector to go into oscillation. If the feedback can be reduced quickly enough during this interval of time, oscillation can be prevented. Feedback can be reduced temporarily either by increasing the negative C bias in the grid circuit or by decreasing the plate voltage. Either can be accomplished by introducing into the regenerative detector circuit a varying voltage, which may be sinusoidal in wave form, and which is known as the quenching signal. The quenching frequency must be somewhat higher than the intelligence signal frequencies; otherwise, it would create interference. For code and speech reception, a quenching frequency of between 10 and 20 kc. is quite satisfactory.

A super-regenerative detector circuit, in which the tube acts both as



a regenerative detector and as an oscillator for producing the quenching frequency, is shown in Fig. 32. Resonant circuits C_2 - L_2 in the grid circuit and C_{3} - L_{3} in the plate circuit act in conjunction with the tube as an oscillator, with feedback provided either by the grid-to-plate capacity of the tube or by mutual induction between coils L_2 and L_3 . The circuit can be tuned to maximum sensitivity without any danger that a carrier increase will send it into audible oscillation. for the inaudible quenching signal always reduces the plate voltage before the tube has time to break into oscillation at the incoming signal frequency.

Of course, a separate oscillator may be used to produce the quenching frequency; this is often done in order to secure independent control of the quenching signal.

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Demodulators for F.M. Receivers

So far, the detectors described have been intended to handle amplitude---- modulated (a.m.) signals. To complete our study of detectors, we must also consider the fundamentals of detectors intended for frequency-modulated (f.m.) signals. Of course, we shall give more information on these later, when we discuss the complete f.m. receiver. For now, we shall present basic circuits so you can compare the detectors used in the two modulating systems. You may be surprised to see how little difference there is between basic circuits-a change in a part value here, or a different operating voltage there, may be all that is needed to make an a.m. stage into an f.m. stage. Again, you will see the value of understanding clearly how parts and circuits work.

► What is the difference between a.m. and f.m.? Essentially, amplitude modulation is a system in which the *amplitude* of an r.f. voltage is varied by an intelligence signal: the *ampli*tude variation being proportional to the *amplitude* of the intelligence signal, and the rate of variation being proportional to the *frequency* of the intelligence signal.

In frequency modulation, the amplitude of the r.f. signal remains constant—the frequency is varied by the intelligence: signal. Here, the frequency variation is proportional to the amplitude of the intelligence signal (louder sounds produce a wider frequency change), while the rate of frequency variation is proportional to the frequency of the intelligence signal.

Thus, in a.m., the *amplitude* changes of the intelligence signal produce *amplitude* changes in the carrier; while in f.m. the *amplitude* changes of the intelligence signal produce frequency changes in the carrier.

A tube is an amplitude-following device—the larger the input signal amplitude, the greater the resulting current. Hence, a.m. detectors produce an intelligence signal output that varies directly with the amplitude changes of the incoming r.f. signal.

On the other hand, the f.m. wave has a constant amplitude: there is no way in which a *tube* can produce an intelligence signal with a varying amplitude from such an r.f. signal. (The tube responds equally well to all r.f. signals over a wide range, as its output is relatively independent of r.f. frequency.) Therefore, it is necessary to add another step to the process of f.m. demodulation. WE MUST FIRST CONVERT THE F.M. SIGNAL TO ONE WITH A VARYING AMPLI-TUDE BEFORE WE CAN DE-MODULATE THE SIGNAL. This is the fundamental difference between a.m. and f.m. demodulators. Let's see how the conversion is made.

A BASIC F.M. DEMODULATOR

To convert an f.m. wave into one having a.m. characteristics, we must have a circuit or device that can discriminate between frequencies. That is, the circuit must give different output voltages for different frequencies. (As discrimination is a necessary step in f.m. detection, the f.m. detector stage is commonly called a discriminator stage.)

Frequency discrimination is obtained by the use of L-C resonant circuits. Let's look at several systems of discrimination, as differences in the methods of discrimination are the major points of difference between f.m. demodulators.

► Let's suppose we have an L-C circuit with a resonance curve like that shown in Fig. 33. Notice that the L-C values of this circuit cause best response to the resonant frequency. which is 100 kc. It responds less well to frequencies of 90 kc. and 110 kc., and rather poorly to frequencies of 80 kc, and 120 kc.

▶ In an a.m. detector, we tune the



FIG. 33. A response curve for a resonant circuit.

L-C circuit to the frequency of the carrier in which we are interested; we then get maximum response from this frequency while nearby carriers are rejected. We depend on the modulation to give the varying amplitude signal needed for the proper operation of the detector.

The "carrier" in an f.m. system is called the "resting frequency." This is the frequency transmitted when there is no modulation. When modulation occurs, the frequency of the transmitted wave varies above and below the resting frequency. Now, if we feed an f.m. signal into this same tuned circuit, and the circuit is tuned exactly to the resting frequency, then we will not get the desired results. As the modulation swings the frequency up and down, we will get a variation in amplitude, but it will be always in the same direction—downward.

For example, if the resting frequency is 100 kc., and it is applied to an L-C circuit tuned to this frequency, it will get the maximum resonance step-up. When the signal frequency swings up to 110 kc., we will get less output (see Fig. 33). On the other hand, when it swings down to 90 kc. we will also get less output. Therefore, we are not reconstructing the desired a.c. cycle; we want the output to go down when the frequency changes in one direction, and to go up when it swings in the other direction.

▶ However, we can use this same circuit if we tune it to resonance at a frequency higher than the resting frequency, as shown in Fig. 34. Under these conditions, the output for the 100-kc. resting frequency is not as great as the output for the 110-kc. signal. On the other hand, the output for 100 kc. is greater than the output for 90 kc. Therefore, variations in frequency above and below the resting frequency will give us variations in amplitude.

For example, when the incoming frequency increases toward 110 kc.,





we will get a greater output from the resonant circuit and hence a greater voltage from the detector. On the next half-cycle, when the incoming signal swings downward in frequency towards 90 kc., we will have a smaller input to the detector, and thus obtain a reduced output. Thus, as the frequency changes above and below the resting frequency, we obtain a varying amplitude signal. Also, the greater the frequency change from the resting value, the greater the amplitude variation and, hence, the greater the output. In other words, an ordinary resonant circuit that is *not* tuned to the resting frequency will_convert an f.m. wave into an a.m. wave, which we can then demodulate in the usual manner.

Actually, a simple f.m. demodulator could be made from any diode circuit just by changing the resonant point of the tuned circuit at its input. However, this is not the system used



FIG. 35. An f.m. discriminator that used two resonant circuits.

in commercial receivers. There are alignment problems, and it is difficult to design a resonant circuit with a response curve that is straight enough to give freedom from distortion. Let's go on and look at commercial systems.

DUAL L-C SYSTEM

Fig. 35A shows one commercial adaptation of the principles we have just described. The circuit works as follows:

Resonant circuit L_1 - C_1 is tuned to

the i.f. resting signal, but it is designed with a low Q so that it is sufficiently broad to respond well to frequencies on both sides of this value. This circuit induces voltages in both L_{2} and L_{3} ;-these voltages cause electrons to flow through diodes VT_{1} and VT_{2} .

Since electrons move only from the cathode to the plate, electrons flowing through VT_1 come "upward" through resistor R_1 , producing a voltage drop across this resistor with the polarity shown. Similarly, electrons flowing through VT_2 move "downward" through VT_2 move "downward" through resistor R_2 . The voltage drop across this resistor is, therefore, of opposite polarity to that across resistor R_1 .

The resonant circuit M (L_2-C_2) is tuned to a frequency somewhat lower than that of L_1-C_1 , while N (L_3-C_8) is tuned to a frequency somewhat higher. If the frequency of 4 mc. in Fig. 35B represents the L_1-C_1 resonant frequency (the resting frequency), then 3.9 mc. may represent the resonant frequency of M and 4.1 mc. that of N. Hence, neither M nor N is tuned to favor the center frequency, although neither excludes signals of this frequency altogether.

The circuits are adjusted so that, at the resting frequency, the voltages applied to VT_1 and VT_2 are equal, and equal currents flow through R_1 and R_2 . As these are equal resistances, their voltage drops are equal under these conditions. Further, since their voltage drops are opposite in polarity (see Fig. 35A), there is no net voltage between terminals X and Y when only the resting frequency is fed into the stage.

Now let us suppose the signal frequency is modulated so that it swings from 4 to 3.9, back through 4 to 4.1, then back toward 3.9, etc.

As the frequency approaches 3.9 (the resonant frequency of M — see

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Fig. 35), more voltage will be developed by M. This causes larger voltages to be applied to VT_1 ; larger currents result and produce larger voltages across R_1 . At the same time, the signal frequency is moving away from the resonant frequency of N, so less voltage is being applied to VT_2 . The resulting smaller current produces less voltage across R_2 . Hence, for this swing, the voltage across R_1 increases while that across R_2 decreases. The two voltages no longer cancel one another, so a net voltage (equal to the difference between them) appears between terminals X and Y.

For example, suppose the R_1 and R_2 voltages are initially 50 volts each,

Summary. A signal, shifting in frequency, is introduced into this stage. The frequency variations produce a varying voltage across terminals X and Y. Small frequency changes produce small voltages, as frequencies near 4 mc. do not "climb" as high on the M and N resonance curves. Larger changes, out to the limits set by the resonance points (3.9 and 4.1 in Fig. 35B) produce larger voltages. Thus, we have a circuit that produces an amplitude variation from the varying frequency signal.

This same circuit also gives us demodulation in the same manner as the diode detector we have discussed for amplitude-modulated signals. The



FIG. 36. This phase-operated f.m. discriminator is the type most commonly used.

and cancel exactly. On the swing just described, the R_1 voltage may go up to 90 volts and the R_2 voltage may go down to 10 volts. The difference is 90 minus 10 or 80 volts, which will appear between X and Y. (Terminal X will be positive with respect to Y, as the polarity will be that of the voltage across resistor R_1 .)

When the swing reverses toward 4.1 mc., exactly the reverse action occurs. Now, N develops the larger voltage, VT_2 passes the larger current, and the drop across R_2 is larger than that across R_1 . This reverses the polarity of the difference voltage existing across X and Y, since the polarity now is that of the voltage across R_2 . diodes VT_1 and VT_2 rectify the r.f. signal, while condensers C_4 and C_5 act as r.f. filters. Thus, the original intelligence signal appears across terminals X and Y.

A PHASE-OPERATED DISCRIMINATOR

Fig. 36 shows the most widely used f.m. demodulator. The circuit is very similar to that of a push-pull a.m. detector described earlier in this lesson, except for the arrangement made to obtain frequency discrimination.

Complete details on the operation of this circuit will be given in a later lesson. Briefly, here's how it works:

 L_1 is the primary, and L_2 and L_3 are

the two halves of the secondary, of a center-tapped i.f. transformer. $(L_2$ and L_3 are actually parts of the same secondary coil; we've given them separate names just for convenience in reference.) The circuit L_1 - C_1 is tuned to resonance at the resting frequency, as is also the circuit L_2 - L_3 - C_2 . A choke coil L_4 is connected to the center point of the transformer secondary, and also (through condenser C_3) to the top of the primary L_1 . This coil L_4 is in the common plate return of both diodes.

If you trace the circuit from the top of L_1 through C_3 , L_4 , and C_5 to ground, and from ground through C_6 to the bottom end of L_1 , you will see that L_4 is effectively in parallel with the primary L_1 . (We can ignore the condensers, because their reactances are negligible at f.m. frequencies.) Therefore, the primary voltage across L_1 is also always across L_4 , and we can call the voltage across L_4 the "primary voltage" for convenience.

At any time, the voltage applied to the plate of diode VT_1 is the sum of the voltage across L_2 and the primary voltage across L_4 . Similarly, the voltage applied to the plate of VT_2 is the sum of the voltage across L_3 and the primary voltage across L_4 . These must be *vector* sums, since the voltages across these coils are not in phase.

This circuit is so arranged that when the resting frequency is fed in (that is, the frequency to which the tuned circuits are resonant) the voltage across L_2 is 90° out of phase with the primary voltage. The voltage across L_3 is then also 90° out of phase with the primary voltage, but in the opposite sense. Say we assume that the L_2 voltage leads the primary voltage by 90°; the L_3 voltage will then lag the primary voltage by 90°. The vector sum of the L_2 voltage and the primary voltage will then be of the same magnitude as the vector sum of the L_3 voltage and the primary voltage. Therefore, when the resting frequency is applied to this circuit, the total_voltage_applied_to the plate-of-each diode is the same. That is, the voltage e_1 for VT_1 will equal voltage e_2 for VT_2 . Equal voltages cause the diodes to pass equal currents through R_1 and R_2 . Since the resistances are equal, their voltage drops are equal and opposite, so no net voltage appears between terminals X and Y.

► However, the two plate voltages do not remain the same when a modulated signal is fed in. As you know, a resonant circuit changes its characteristics when a signal other than the resonant frequency signal is applied to it. If an above-resonance frequency is applied, the circuit acts like a coil; if a below-resonance frequency is applied, the circuit acts like a condenser. In other words, the phase of the circuit voltage changes when an offresonance frequency is applied.

When a modulated signal is applied to our discriminator circuit, phase changes occur both in the primary voltage and in the voltages across L_2 and L_3 . The amount of phase change will depend on how far from the resting frequency (how far off-resonance) the modulated signal swings. We might have, say, a total shift of 45°. The L_2 voltage might then lead the primary voltage by 135° (90° plus 45°); the L_3 voltage would then lag the primary voltage by only 45° (90° minus 45°).

Obviously, under these conditions the vector sum of the L_2 voltage and the primary voltage is not equal to the vector sum of the L_3 voltage and the primary voltage. Thus, the phase shift causes unequal plate voltages to be applied to the diodes. Therefore, the R_1 and R_2 currents are unequal and a net voltage appears between terminals X and Y. This voltage becomes larger the farther the f.m. signal swings from the resting frequency, and reverses in polarity as the signal swings above and below the resting frequency. It is, therefore, the amplitude-modulated signal we want. Detection of this a.m. signal occurs as it does in the circuit in Fig. 35; the diodes rectify the r.f., and condensers C_4 and C_5 act as r.f. filters, so the intelligence signal appears across terminals X and Y.

► Don't worry if all the details of these f.m. circuits aren't perfectly clear to you now—we will discuss f.m.

receivers thoroughly in later lessons. Even now, though, you can see that f.m. sets are not tremendously different from a.m. receivers. In fact, except for the addition of the frequency discriminator and the use of a circuit called a limiter, the f.m. receiver is designed very similarly to a wideband high-fidelity a.m. receiver. And, as we have just shown, the f.m. signal must be converted into an a.m. signal at the detector before we can obtain it. The f.m. system of modulation has advantages; notably, high certain fidelity capabilities and, when a limiter stage is used, a remarkable freedom from noise.

Reviewing Methods of Providing C Bias

Now that we have studied nearly all the basic vacuum tube circuits, let us review the different ways of providing the negative operating C bias required in a.f. and r.f. amplifiers, detectors, oscillators—in fact, in any vacuum tube circuit. You are already familiar with most of these methods, but we will review them briefly here to give you a complete picture.

You will recall that a grid bias is a voltage applied to the grid to fix the operating point of the tube at the proper point on the characteristic curve. It is possible to get an entirely different operating action from a radio stage just by varying the bias voltage. For example, you know that a class A amplifier — the kind most commonly used in receiver circuits — has a bias voltage that makes it operate on the straight portion of its characteristic. The bias keeps the grid negative at all times (providing the signal level is not excessive) so it will not draw grid current.

If we increase the bias enough sc that the tube plate current is practically zero (in other words, if we bias the tube almost to cut-off), we have class B operation. We use class B stages in push-pull audio amplifiers, in transmitters, and as detectors.

Increasing the bias still further (well beyond cut-off) will give us class C operation. A class C amplifier is a highly efficient circuit used only for r.f. power amplification and for oscillatory circuits.

► Obviously, the bias voltage is important to the operation of the tube stage. We can secure bias in various ways: we can use a battery, or arrange a self-bias circuit in which plate, filament, or grid current provides the bias voltage;* or we can use fixed

^{*}A.V.C., an automatic C bias dependent on signal level, is not included in this review, but will be described in a later lesson.

bias circuits in which the bias voltage is obtained from the power supply. Any of these methods can be used with any tube which has a grid.

BATTERY BIAS

Fig. 37 shows several methods of supplying a bias voltage from a battery. Exactly the same method is shown in A and B, except that A shows a tube with a filament type cathode, while B shows a tube with a separately heated cathode.

As you see, the C battery is connected in the grid return circuit.



FIG. 37. Several methods of using batteries to furnish bias.

Hence, its voltage appears between the grid and cathode, and thus sets the operating characteristics of the tube.

Since the grid is made negative by the bias voltage, no grid current flows. Therefore, the bias voltage causes no current flow through R_{g} , so this resistor does not enter into the biasing arrangement in these or any succeeding diagrams. The sole purpose of R_{g} is to provide a load across which signal voltage can be developed. (When signals are applied, there will be a signal current through $R_{\rm g}$ that will cause a signal voltage drop across this resistor. This signal voltage will alternately add to and subtract from the bias voltage, swinging the gridvoltage above and below the operating point.)

▶ In Fig. 37C, a bias cell is shown in use. A bias cell is actually a very small chemical-cell battery. Two types of bias cells are available; one type is rated at 1 volt while the other gives $1\frac{1}{4}$ volts. Any reasonable bias voltage can be obtained by placing several in series. However, it is rare to find more than two used in a radio receiver.

The advantages of the bias cell are that it is small in size and that it can be wired into the circuit as a semipermanent part of the circuit, since it does not require frequent replacement. Of course, the cell must be used properly to have reasonable life. Never try to measure the voltage of a bias cell with a voltmeter, because the current drawn by the meter may damage the cell. If you suspect that a bias cell is defective, remove it and substitute another, or a battery, in its place. You can determine whether the original is defective by observing the operation of the stage when the substitute is being used.

SELF-BIAS METHODS

There are a number of biasing methods that depend upon plate, filament, or grid current flow. Let's take each up in turn.

Plate Current Methods. Fig. 38 shows several methods whereby the average d.c. plate current flow through a resistor produces a voltage that can be used for bias purposes.

In Fig. 38A, the plate current flows through resistor R_c . (Electrons flow from the cathode to the plate, through

 $R_{\rm L}$, through the B supply, and through $R_{\rm c}$ back to the cathode.) As a result, a voltage drop that has the polarity shown in this figure exists across $R_{\rm c}$. Since resistor $R_{\rm c}$ is between the grid and cathode, the voltage across it is also between the grid and cathode. This voltage makes the grid negative.

We can adjust this bias to the required amount by choosing the proper value of resistance for the plate current of the tube.

The condenser $C_{\rm c}$ by-passes the bias resistor. This prevents any of the a.c. signal voltage from being lost in $R_{\rm c}$, and also prevents the plate circuit a.c. voltage from causing degeneration in the grid circuit. which the plate current flow passes to the filament. In this case, remember that the current is doubled, since it comes from two tubes. As a result, the bias resistor should be half the size that is necessary for a single tube to produce the same bias voltage.

Filament Current Methods. Fig. 39 shows how the filament current can be used with battery type tubes to supply a bias. (We cannot use the filament current of a.c.-operated tubes as a bias source, because the filament current variations would introduce a 60-cycle variation in the plate current, producing hum. We cannot permit variations in the bias, so we can get bias only from a d.c. source.)



FIG. 38. Typical self-bias circuits.

► Fig. 38B shows the same kind of circuit for a tube using the filament as the cathode. It operates precisely the same way.

Fig. 38C shows a similar arrangement for a tube with an a.c.-operated filament type cathode. In this case, there is a center tap on the filament winding. The plate electrons flow from the cathode to the plate, through the primary of transformer T_A , through the B supply, and through R_c to the center tap on the filament winding, then back to the filament.

▶ 38D shows the same arrangement for a push-pull stage. Here the plate currents of both tubes flow through R_c to the center-tapped resistor R, from In Fig. 39A, electrons circulate from the battery through resistor $R_{\rm F}$, through the filament, and back to the battery.

► The amount of bias depends on the reference point used. There are two systems of reference points used with these battery type tubes. As you know, there is a voltage drop along the filament wire, caused by its resistance, so that different points on the filament are at differing potentials with respect to the grid. Thus, the grid is far more negative with respect to the filament than it is with respect to the other.

Formerly, the middle of the filament was taken as the bias reference point. Following this system, the bias voltage in Fig. 39.4 will be half the filament voltage drop, plus the drop across the resistor $R_{\rm F}$. This adds up to a bias of 3.5 volts. However, modern practice has been to assume that the negative side of the filament is the reference point, in which case the 1-volt drop across resistor $R_{\rm F}$ is the bias voltage.

▶ In Fig. 39B, we have a filament string so arranged that the drops



made to supply bias.

across the tube filaments become bias voltages. Let's see how this circuit works. We will consider the negative filament terminals to be the bias reference points.

The filament electrons move from A-, through the resistor $R_{\rm F}$, through the filament of tube VT_1 then through the filaments of VT_2 and VT_3 , to the A+ terminal. As far as tube VT_1 is concerned, the only bias voltage between its negative terminal (terminal 1) and its grid is the drop across the resistor, $R_{\rm F}$. This, then, is the bias for VT_1 .

Now let's look at VT_2 . The voltage between its grid circuit and point 2on the filament string includes both the drop across R_F and that across the filament of tube VT_1 . Therefore, this tube gets a higher bias, because its grid resistor R_{r2} returns to the point 0 on the filament string.

Similarly, tube VT_3 has a bias equal to the drop across R_F plus the drops across the filament of VT_1 and the filament of VT_2 . In other words, the bias for this tube is equal to the sum of the drops between the return point of its grid resistor R_{g3} (point 0) and point 3 on the filament string.

Thus, by including more and more of the series filament voltage drops, the bias voltage on succeeding tubes can be increased. Or, we can adjust the bias by returning the grid resistor of each tube to the desired point on the string.

Grid Current Methods. You are already familiar with the fact that grid current flow through a grid leak resistor will provide a bias voltage. The method is shown in Fig. 40A. Whether the grid resistor is at R_{c1} or at R_{c2} , the same effect is obtained. In either case, the signal voltage swings the grid positive to a sufficient extent for grid current to flow. The



FIG. 40. The grid current can be used for bias purposes, as shown here.

grid current flow through the resistor then provides the bias voltage.

A modern adaptation of this method is shown in Fig. 40B. Here, resistor R_c appears to be the usual grid resistor, used only as the load across which we develop a signal voltage. However, if we make this grid resistor very high in value, the tiny amount of convection current that normally flows through any tube will set up a small bias voltage. (Convection current is the small grid current caused by some of the electrons speeding toward the plate happening to strike the grid wires. As these electrons are caught, they make the grid negative with respect to the cathode, so they tend to drift back to the more positive cathode through the grid circuit.) For some tubes, particularly certain first audio tubes, the bias developed in this manner is sufficient for normal operation.

Notice the difference between the eircuits in Figs. 40A and 40B. In B, the bias is due to stray electrons,

vantage, but not always; in fact, in some circuits self-bias is undesirable for precisely this reason. There are other circuits in which it is impossible to use self-bias. For such circuits, we must use fixed bias sources—either batteries, which we have already discussed, or the set power supply. Fig. 41 shows several different ways in which a bias voltage can be obtained from the power supply.

In Fig. 41A, we have a combination of self and fixed bias. The plate current of the tube flows through R_8 which provides a certain amount of self biasing. In addition, there is an



FIG. 41. When a fixed bias is needed, it can be obtained from the power supply by any of these methods.

which are captured by the grid and cause an extremely small grid current flow through a very high resistance. (R_o may be 10 to 20 megohms.) In this case, the tube has a bias whether or not a signal is applied. In Fig. 40A, however, bias exists only when a signal is applied.

FIXED BIAS METHODS

Most of the self-bias methods are self-regulating—that is, the bias goes up when the tube current increases, and down when it decreases, and so tends to correct for variations in tube characteristics. This is usually an ad-

electron path from B— through R_3 , bleeder resistor R_2 , and the screen supply resistor R_1 to B+. This extra bleeder current makes the current flow through R_3 higher than the plate current value and increases the voltage drop across it. Thus, we can get the required bias with a smaller value of resistor R_3 . Also, if we adjust the bleeder resistor R_2 properly, we can make the bias voltage depend primarily on the bleeder current rather than on the tube plate current. The bias then becomes at least semi-fixed. as it does not depend so much on the tube characteristics.

Fig. 41B shows another combination of self and fixed bias. The plate currents of all the tubes flow through R_1 . The resulting voltage drop is used to bias VT.

▶ Fig. 41C is similar to Fig. 41B, except for the use of L instead of R_{1} . In this case, the filter choke (or the speaker field being used as a choke) is in the negative side of the power supply. The drop across its d.c. resistance becomes the bias for VT. As there is also a large a.c. drop across the inductance L, a filter R_2 -C₁ is used to prevent the a.c. components from reaching the grid.

▶ When the speaker field is used as a choke, the drop across it is usually larger than is required for bias purposes. In some cases a tap is placed on the field so that a portion of the drop may be used. The connections will then be similar to those shown in Fig. 41C, except that R_1 will go to the tap of L instead of its B— terminal.

Another arrangement uses a voltage divider like that shown in Fig. 41D. Resistors R_1 and R_2 divide the d.c. voltage drop across L, and the portion of the voltage across R_2 becomes the bias. The amount of bias can be adjusted by properly choosing the values of R_1 and R_2 . These resistors have high values so they will not reduce the effectiveness of L as a filter. (High values are possible because no d.c. current flows in the grid circuit.) As you will notice, most of the fixed bias methods also are self bias systems to a certain extent. However, if the bleeder current, or the current from the other tubes, is a large portion of the current through the bias resistor (or coil), then the bias will be relatively fixed.

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Lesson Questions

Be sure to number your Answer Sheet 17FR-3.

Place your Student Number on every Answer Sheet.

Send in your set of answers for this lesson immediately after you finish them, as instructed in the Study Schedule. This will give you the greatest possible benefit from our speedy personal grading service.

- 1. What are the two steps involved in the demodulation (detection) of an a.m. signal?
- 2. What are the three performance characteristics on which detectors are rated?
- 3. Does a diode give better fidelity on: 1, strong signals; or 2, weak signals? \downarrow
- 4. What would happen to the higher audio frequencies if the value of the V detector load resistance is increased (in a diode detector circuit)?
- 5. What use is made of the d.c. component of the load voltage developed in diode detector circuits?
- 6. Across which one of the following condensers in Fig. 21 is there neither r.f. nor a.f. voltage: C₂; C₃; C₅?
- 7. In the television detector shown in Fig. 22, what furnishes the filtering capacity?
- 8. In a triode grid-leak detector, does rectification occur: 1, in the plate circuit; or 2, in the grid circuit?
- 9. What step, prior to rectification and signal separation, must be performed in an f.m. demodulator?
- 10. How does the tube in Fig. 40B get its C bias?

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ACHIEVEMENT BRINGS SATISFACTION

The more difficult the achievement, the greater is the satisfaction we feel. In the words of the French writer Moliere, "The greater the obstacle, the more glory we have in overcoming it."

Each achievement gives us the confidence to tackle still larger jobs and overcome greater obstacles. For instance, if this particular lesson is harder for you to understand than previous lessons, but you stick with it until you've mastered the important facts, you'll feel pretty good about completing it and you'll start the next lesson with real confidence in your own ability.

My greatest satisfaction today comes from teaching others how to achieve success, so your achievements mean a lot to me.

J. E. SMITH



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STUDY SCHEDULE NO. 18

A bird's eye view of the rapid growth of radio, from the crystal set of three decades ago to the superheterodyne receiver of today, is presented in this book. Stirring history-making events like the sending of the first wireless signals across the Atlantic are interwoven with achievements of men like Marconi, De Forest, Armstrong, Hertz and hundreds of others. You will learn how each new receiver evolved logically from the preceding, and how practical requirements of the listening public resulted in one improvement after another.

This book is in its present position in your Course because a knowledge of the technical problems which have been solved by the pioneers in radio will help you to understand modern circuits. Also, some of these older receivers are still in use; the information given will be of real practical value when they require servicing, and will tell you what to expect from these older sets in the way of performance.

	١.	Radio Wireless
Í	2.	Radio Broadcasting is Born
ľ	3.	Receivers Grow Up
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M	5.	Modern Receiver Types
V	6.	Mail Your Answers for this Lesson to N.R.I. for Grading.
	7.	Study One Circuit in Reference Text 17X. To get the most benefit from this valuable and practical reference text, study one circuit in it per lesson. One slow and careful reading will do, if you look at the diagram each time a part is mentioned. Do this for all the circuits, even though they may not be mentioned in the Study Schedules of other lessons.
	8.	Start Studying the Next Lesson.

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World Radio History

HOW PRACTICAL REQUIREMENTS CHANGED RECEIVER CIRCUITS

Radio Wireless

AFTER the first novelty of owning a radio wore off, back in the days when the radio industry was just getting started, owners of sets began to want something better. Receiver designers learned of these requirements and undertook research with a view to improving the receivers. New discoveries resulted and were quickly adopted; innovation followed innovation as consumers kept demanding better receivers.

► The circuit chosen for any particular receiver depends upon seven important technical requirements demanded by the public for that type of receiver:

- 1. selectivity;
- 2. sensitivity;
- 3. fidelity;
- 4. power output;
- 5. signal-to-noise ratio;
- 6. interference reduction;
- 7. ease of operation.

Today the radio engineer can produce receivers which meet the requirements of the public, but it took a long time to reach this stage. Early radio receivers were of decidedly limited ability.

So that you can better appreciate the technical perfection of the presentday radio receiver, as well as realize why various receiver models today differ so greatly from each other, we shall trace the development of radio receivers from the days prior to the crystal detector to the all-wave, highfidelity superheterodyne receivers of today.

COHERERS TO DIODE DETECTORS

About 1864 James Clerk Maxwell. one of the great British scientists, predicted by means of complicated mathematics that an oscillating current would radiate electromagnetic waves which would be identical with light waves except for frequency. These predictions aroused considerable scientific argument. For twenty vears Heinrich Hertz, a German scientist, attempted to prove that Maxwell's assumptions were wrong and that these electromagnetic waves could not exist. The results of these experiments quite surprised Hertz, finally converting him to the Maxwell theory. As a result, Hertz published in 1887 and 1889 the papers which brought . him fame and gave to the world experimental proof that radio waves were entirely possible. Even today radio waves are designated by many people as Hertzian waves.

In 1890 Sir Oliver Lodge, another brilliant British scientist, actually set up an oscillating circuit (a charged condenser shunted by a heavy wire loop in series with a spark gap), and proved that another oscillatory circuit would pick up by induction the energy from the first circuit, if properly tuned.

The Coherer. The next important development before the actual sending of messages through the air was the coherer, a device which is based upon the fact that the enormous resistance offered to the passage of electric current by loose metal filings is greatly reduced under the influence of alternating current.

In one very popular form, a coherer consisted of a two- or three-inch long glass tube, inside which were metal plugs spaced perhaps a quarter inch apart, with the space between the plugs_ filled with metallic filings. Various metals were used for the plugs or electrodes and for the filings. The passage of radio frequency current through the filings caused them to "cohere" or stick together, forming a conducting path for the direct current which actuated a telegraph sounder (an electromagnet which makes an audible click each time it operates). The great problem was to break up this path again after the r.f.

1896 to combine the results of all these scientists and produce one simple and workable radio system. His early apparatus was capable of transmitting messages for only short distances, but on December 12, 1901 he used the transmitter and receiver shown in Fig. 1 to transmit three dots (the letter S) over and over again from Poldhu, England to St. Johns, Newfoundland. To Marconi rightfully belongs the title "father of wireless," for he coordinated the work of others and put wireless on a practical working basis. (Before the advent of broadcasting, radio was known as wireless.)

 \blacktriangleright In order to secure long-distance transmission in those days, it was



FIG. 1. Early Marconi wireless transmitting and receiving circuits. The vibrator at the transmitter operated as long as the key was pressed, interrupting the primary current of the induction coil and inducing in the secondary a high voltage. This high voltage caused an oscillatory spark discharge across the spark gap. The oscillations were radiated into space by the transmitting entenna, and induced r.f. currents in the receiving antenna. These currents made the coherer conductive, so that battery current passed through to operate the tapper and sounder. The buzzing sound of the tapper (an ordinary buzzer) was thus heard whenever the sending key was pressed.

current had ceased flowing; some experimenters mounted the coherer on a vibrating table, but the most popular arrangement was that used by Popoff of Russia and by Branly, in which a tapper attached to the telegraph sounder was used to jar the glass tube after the arrival of each signal.

► Scores of scientific men began experimenting with the newly-discovered Hertzian waves, each contributing additional data on the behavior of these waves, but it remained for Marconi in

necessary to use tremendously highpower oscillatory circuits, together with antennas much longer than those in common use today.

Having increased the power of transmitters to the practical limits of the time, scientists now turned their attention to improvements in receivers. The electrolytic detector, a fine platinum wire immersed in sulphuric acid, came into the picture around 1903. It was about this time also that the headphone unit used in telephone systems was brought into play to replace the telegraph sounder used for wireless. This one step increased the sensitivity of receivers considerably.

Crystal Detectors. At about this same time the crystal came into widespread use as a more convenient and



FIG. 2. A cross-section view of a typical crystal detector assembly is shown at A, and a typical crystal set diagram is shown at B.

more sensitive detector. Way back in 1874. Braun had discovered that two crystals making point-to-point contact with each other had a high resistance to the flow of current in one direction and a low resistance to current flow in the opposite direction. Single crystals in contact with a sharpened length of fine wire were found to give equally good rectifying action; materials such as galena, iron pyrite, fused silicon, carborundum, molybdenite, and other minerals came into use. A typical crystal detector is shown in Fig. 2A; this has the necessary mechanical adjustment which permits setting the cat'swhisker at a spot on the crystal which gives good rectification.

With the development of more sen-

sitive crystal detectors, together with sensitive headphones, long-distance reception became possible, and more and more transmitters went on the air. These were all code or telegraph transmitters, of course, and were operating on wavelengths longer than 600 meters (lower than 500 kc.). Congestion of the airways became severe, and steps were taken to make transmitters use. narrower bands of frequencies and to make receivers more selective.

One of the earliest schemes for improving the selectivity (station-separating ability) of a receiver was the use of variable coupling between the antenna coil and the secondary coil, as illustrated in Fig. 2B. Additional coils, known as *loading coils*, were sometimes inserted in the tuning circuits at points x and y to increase the circuit inductance and thus give reception on the longer wavelengths (lower frequencies).

The Diode Tube. In 1904 Dr. Fleming invented the diode vacuum tube, then known as the *Fleming valve*.



FIG. 3. An early diode tube detector circuit.

This contained a wire filament heated by direct current, and a metal plate or anode mounted a short distance away, both electrodes being enclosed in an evacuated glass envelope. A typical Fleming valve detector circuit is shown in Fig. 3. The same tuning controls are used here as in the crystal detector circuit of Fig. 2B to secure the required selectivity. Actually the two circuits are identical except for the type of detector used.

DE FOREST INTRODUCES TRIODE TUBES

The introduction of the grid in the Fleming diode tuhe by Leo De Forestin 1909 was an outstanding contribution to the radio art, and revolutionized the entire radio industry. The De Forest triode tube was first used only



FIG. 4. One of the first triode tube detector circuits.

in detector circuits, where it provided a considerable amount of signal amplification in addition to demodulation.

A typical triode detector circuit is shown in Fig. 4. The output of the tuned circuit L_{T} - C_{T} was fed to the grid and filament of the triode tube through series condenser C_{s} . The headphones, in series with a 221/2- to 45-volt battery, were connected between the plate and filament of the tube. It was found that a leaky condenser gave best results for $C_{\rm g}$ (the reason for this was not at first apparent, but is now fully understood — the intelligence signal voltage was developed across the resistance of this leaky condenser and then amplified by the triode tube). Thus the filament and grid of the tube together acted as a diode rectifier, while the entire tube acted as a lowfrequency amplifier. This circuit (known as a grid-leak and condenser detector) is still used to a certain extent.

Early triode tubes were rather poorly evacuated and therefore contained a certain amount of gas. This gas made possible a larger plate current than would exist in a pure vacuum. These gas tubes were commonly known as "soft" tubes and made very sensitive detectors. However, they were difficult to adjust properly as detectors, they were noisy, and their useful life was very short. Improving the vacuum in the tube (giving a so-called "hard" tube) gave better stability and longer life, but somewhat poorer sensitivity.

The loss in sensitivity through the use of a hard tube was more than overcome by the invention of regeneration by De Forest in 1912 and by Armstrong in 1914. A simplified form of this regenerative circuit is shown in Fig. 5; here the modulated r.f. current





which exists in the plate circuit is fed back to the input circuit to reinforce the incoming signal. In the hands of an expert the regenerative detector gave amazing results, setting up new and more consistent long-distance receiving records.

AUDIO AMPLIFICATION

Attempts to bring up the volume of very weak code signals by increasing the feedback of a regenerative detector were decidedly unsatisfactory, for the



receiver of 1920-1924 vintage, having two a.f. stages. These receivers had witch SW; separate filament controls R_1 , R_2 and R_2 ; a tuning condenser; a regeneraontrol varying the coupling between T_1 and T_3 ; variable coupling between T_2 and T_3 and sometimes a tapped loading coil arrangement like T_4 to extend the tuning range. Thus, there were seven or eight controls on the panel of these receivers. When receiving strong signals, phones could be plugged into jack J_1 , which disconnected the a.f. tube filaments, prolonging both tube and battery life. With the phones plugged into J_2 , only the first a.f. stage was in use, while full amplification for weak signals was obtained with the phones in J_3 . The first horn speakers could be plugged into J_3 also.

circuit became unstable and often went into oscillation, producing very annoying squeals. Practical requirements demanded amplification of weak but clearly demodulated signals — audio amplification.

As soon as suitable tubes were available, one or more stages of audio amplification were added to the regenerative detector to increase the volume of weak code signals. Plug-in jacks were provided for each stage, as in the typical circuit shown in Fig. 6, so that headphones could be inserted either in the detector stage, the first audio stage or the second audio stage. This was a crude but effective means of controlling volume. Although this was a great improvement over any other receiver available at the time, manufacturers of radio apparatus were not yet ready to scrap the then widely used crystal detectors; they simply manufactured a separate two-stage audio amplifier which could be connected to the output of any crystal detector.

Radio Broadcasting is Born

With code wireless an established fact, engineers turned their attention to the transmission of voice and music through the air. The old spark coil transmitter, with its highly damped oscillations, was clearly inadequate for the transmission of voice signals. In 1903 Poulsen had perfected a generator of high-frequency current which utilized an electric arc; he showed experimentally that telephonic communication through space over very short

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distances was entirely possible with his transmitter if a microphone were used to modulate the radio waves. The Poulsen arc transmitter was used extensively for code transmission, but did not prove entirely satisfactory for voice signals.

► A high-frequency dynamo-electric generator was designed and constructed by Fessenden in 1906 for the production of continuous waves; with it he was able to transmit telephone messages for 15 miles.

► De Forest in 1907 was the first to use wireless for the broadcasting of phonograph records; he placed his musical program on the air without



FIG. 7. A photograph of an early receiver.

preliminary announcement, completely surprising the wireless code operators on duty at the time.

► The invention of the triode vacuum tube and its development as an oscillator for the generation of high-frequency current resulted in the worldfamous demonstration on December 30, 1915, when the government radio station at Arlington, Virginia, succeeded in sending telephone messages through the air to Honolulu, Hawaii and Paris, France.

► During World War I (1914-1918) the large electrical manufacturers in this country had developed elaborate wireless research and manufacturing departments for the production of radio apparatus for the government. The close of the war marked the collapse of the market for their products, and naturally these manufacturers tried to find new ways of stimulating wireless activities and creating new markets. The Westinghouse Electric and Manufacturing Company, believing that the future of radio was in the direction of voice and music broadcasting, delegated Frank Conrad to carry out research work along this line. Fortified with a vast amount of wireless experience, the latest scientific aids, and practically unlimited manufacturing facilities, Conrad developed the first broadcasting station. For months radio amateurs in the vicinity of Pittsburgh listened with interest to experiments in broadcasting, and then, on November 2, 1920, Westinghouse Radio Station KDKA astonished the world by broadcasting the results of the presidential election.

Wireless gradually came to be known as radio, and every one wanted to listen in on the broadcasts. Shortly after this epoch-making broadcast, Westinghouse opened Radio Station WJZ in New York City, the first commercial broadcasting station. Station after station sprang into existence broadcasting swept the country, expanding into a vast industry and proving correct the early prophecies of Westinghouse.

► The nuisance of having to wear headphones whenever listening to radio broadcasts had long been recognized as a distinct drawback to the universal popularity of radio. In 1921 a megaphone was attached to an electromagnetic headphone unit, forming the first practical loudspeaker. The vibrating diaphragm of the phone unit, acting in conjunction with the outward flaring curve of the megaphone, served to excite the surrounding air waves so that all the persons in a room could



Broadcasting the election returns over KDKA on November 2, 1920. This historic picture shows the station engineer, announcer, two newsmen, and the transmitter.

listen to a single program. The immediate acceptance of this rather crude loudspeaker quickly resulted in the development of the balanced armature unit which gave greater diaphragm movement and consequently greater sound output.

During the period between 1922 and 1924, the receiver circuit shown in Fig. 5, used with batteries and a horn-type loudspeaker, was the accepted standard. At this stage in the development of radio, people wanted plenty of "gadgets" on their receivers; the more controls a receiver had, the better it was received by the public. A typical receiver is shown in Fig. 7.

Another form of regenerative receiver was developed about this time. This circuit, known as the super-regenerative, had a special means of preventing oscillation, so the set could operate at the point of maximum amplification. However, the circuit had extremely poor selectivity. It tuned so broadly it could not be used in crowded wavebands and thus lost favor.

T.R.F. RECEIVERS

Regenerative receivers had one very serious fault; they would feed energy



FIG. 8. Circuit of a typical five-tube neutrodyne receiver, the Freed-Eismen model NR-5, which appeared in 1925. Feedback in the r.f. steges was cancelled out by adjusting the neutralizing condensers merked N.C.

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FIG. 9. The t.r.t. amplifier and detector section of a receiver which was popular about 1926. The grid leak and condenser detector was usually followed by two stages of a.f. amplification.

back into the receiving antenna, and this would be radiated, interfering with reception of signals by other receivers in the immediate neighborhood. Owners of radio receivers did not enjoy listening to ear-splitting squeals each time their neighbors decided to tune in a new station, and thus the public created a demand for the next improvement.

Around 1925 engineers seriously began to consider replacing regenerative circuits with circuits using r.f. amplification ahead of a non-regenerative detector. The big problem was making the then available vacuum tubes amplify r.f. signals without causing feedback which would result in shrill squeals accompanying the desired program. The neutrodyne circuit, as developed by Hazeltine, effectively prevented r.f. feedback and proved the solution to this problem.

A typical neutrodyne circuit, as used in receivers of this period, is shown in Fig. 8. There were three tuning dials, one for each resonant circuit tuning condenser in the r.f. tube stages and the detector. Other controls included a rheostat for varying the filament current to the detector and two a.f. amplifier tubes, and another rheostat to control the filament current to the r.f. amplifier tubes. Plug-in jacks (omitted for simplicity in the diagram) were inserted in each audio stage to allow the listener to use headphones instead of the less-sensitive loudspeaker on weak distant stations. People still insisted upon being able to stay up into the wee hours of the morning, with headphones "glued" to their ears, trying for coastto-coast reception. The neutrodyne receiver of 1925 was heralded as the marvel of its day.

This "squeal-less" radio receiver captivated the attention of the radio public. Many manufacturers turned over their entire facilities to the production of these receivers under the Hazeltine patent, in order to meet the rapidly growing demand. Other manufacturers developed their own r.f. feedback cancellation systems, all attempting to balance out the plate-to-grid circuit feedback voltage with an equal and opposite voltage.

Some manufacturers attempted to get around the neutrodyne patent by developing tuned radio frequency circuits which had enough loss in each r.f. stage to absorb the feedback energy and prevent squeals. This "losser" method proved fairly successful.

Grid suppressors, connected as shown in Fig. 9, also came into use as cures for feedback troubles in some of the earlier t.r.f. receivers.

► Other circuit improvements inaugurated at about this period included a variable resistor connected across the primary of the second tuned r.f. transformer for the purpose of controlling regeneration and volume. Fixed resistors and ballast resistors (such as the Amperite) in the filament circuits made it unnecessary to adjust filament current. Three tuning controls and a volume control now were all the adjustinents used for a standard receiver.

The schematic circuit diagram of a tuned radio frequency amplifier using a grid-leak-condenser detector is shown in Fig. 9. Receivers of this type were designed to regenerate slightly when resistor R was set at its highest ohmic value, thus giving perfect control and high r.f. gain.

evident, and the popularity of homemade receivers began to fade.

Out of this period of home-made receivers there stands out one receiver which gave exceptionally good performance. This receiver contained a neutralized t.r.f. amplifier ahead of a regenerative detector, and was commonly known to the old-timers as the Browning-Drake receiver, after its inventors. The circuit diagram for this receiver is given in Fig. 10; one stage of audio applification is shown here, but ordinarily a power stage was also used to provide loudspeaker operation. Home set-building died a natural



HOME-MADE RECEIVERS

From the very first days of wireless, the building of radio receivers in homes was a popular hobby. When radio broadcasting began, many more thousands joined the ranks of receiving set builders. Sets were first built from wiring diagrams, but soon complete kits of parts were made available. In those days just as many radio receivers were home-made as were produced by manufacturers. Gradually, however, as manufacturing techniques were placed on a production basis, the economy and superiority of commercial sets became

FIG. 10. One form of the famous Browning-Drake receiver circuit, the hit of the 1925-1927 period of home-made receivers.

death soon after the introduction of the superheterodyne, primarily because of the increasing complexity of receiver circuits and the necessity of having special instruments for aligning the superheterodyne. Today, people who build their own receivers usually buy complete kits of parts, but the number of these experimenters is relatively small because little or no money can be saved. Modern radio factories, buying parts in very large quantities and using mass production techniques, can assemble complete receivers for less than the price which an experimenter would have to pay for the parts alone.

Receivers Grow Up

The advent of t.r.f. receivers of the neutrodyne and grid suppressor types changed radio from a fad to an established industry, but prior to 1926 it was still looked upon as a fall-winterspring affair. The radio industry struggled and worried through the summer months, wondering if business would return again in the fall. It did come back, bigger and better than ever each fall; with confidence established, business men began to invest great sums of money in radio. These investors could see no real reason for the summer



FIG. 11. Types of tuning dials in common use during the period when radio was passing from multi-dial to single-dial control.

slump, and demanded an organized effort to bridge the gap. Broadcasting stations boosted their power during the summer months, better and better programs were put on the air, and reasons for the summer slump in radio popularity were studied.

SIMPLIFIED TUNING

It was found that many people objected to three-dial receivers on the basis that they had only two hands with which to tune. It was found that two of the three tuning circuits could

be made to operate exactly alike; referring to Figs. 8 and 9 where the tuning circuits are marked 1, 2 and 3, manufacturers designed circuits 2 and 3 to tune exactly alike, permitting control of the two tuning condensers with a single dial. Naturally, to increase ease of tuning, it was desirable to have the dial of circuit 1 (the input tuning circuit) have the same reading as the single dial used for circuits 2 and 3 when the set was tuned to a station. But the effects of the antenna on circuit 1 were quite marked—if the input circuit were designed for a short antenna (S.A.), then the connection of a long antenna (L.A) would make the dial reading for tuning condensers 2 and 3 differ as much as ten dial divisions from that for tuning condenser 1. The difference in the characteristics of long and short antennas was overcome by providing two antenna posts, one for a long antenna and one for a short antenna; with the antenna connected to the proper post, and condensers 2 and 3 ganged together, a two-dial receiver was obtained. One form of twodial control is illustrated in Fig. 11A; the third condenser was attached to the shaft of one of the two condensers shown.

The next step in the simplification of tuning was the introduction of drum dials, which permitted adjusting the two tuning controls either separately or together. The face of one of these two-drum controls is shown in Fig. 11C; one drum controlled the condensers for tuning circuits 2 and 3, and the other controlled the single antenna tuning condenser. Some receivers used three separate drum dials, one for each condenser, while others had all three condensers controlled by a single dial, like that shown in Fig. 11B, for rough



FIG. 12. Two forms of untuned antenna circuits used with t.r.f. receivers.

tuning to a station, and had some means for making fine (vernier) adjustments of each condenser when more accurate tuning was desired.

The ideal single-dial receiver was still sought by radio designers and by the public, for with it even a child would be able to tune in radio broadcasts correctly and easily. It was fully realized that the antenna circuit was the trouble-maker. Untuned antenna circuits like those shown in Figs. 12A and 12B, followed by two stages of tuned r.f. amplification (three variable condenser circuits), were tried but found unsatisfactory. The trouble was that in the vicinity of powerful local stations, the strong r.f. input signal reaching the first r.f. amplifier tube would make that tube work as a detector, and the strong incoming local signal would mix with or modulate whatever other incoming signal was being tuned in. This so-called crossmodulation interference effect was so objectionable that the idea of an untuned antenna system was dropped completely. Incidentally, cross-modulation is still a problem in radio receiver design.

► Engineers reasoned that if the r.f. section could be designed with plenty of gain, exact tuning in the antenna circuit would not be necessary and single-dial controls could be used. Tests proved this reasoning to be correct as far as local and near-distant stations were concerned, and the single-dial receiver became a reality.

The circuit of one of the first popular single-dial receivers is shown in Fig. 13. Notice that it contains a three-stage r.f. amplifier (four ganged variable condensers), a grid-leak-condenser detector and two transformercoupled audio amplifier stages. The variable inductance, then known as a *variometer*, was coupled to the antenna coil; this variometer could be adjusted to tune the antenna exactly to resonance when a weak distant station was being received, but ordinarily was not





FIG. 14. Two methods of tuning the antenna input circuit to resonance with the other circuits. when tuning in weak distant stations.

needed or used when tuning in local stations. There were several taps on the antenna coil for adjusting the circuit to various lengths of antennas; the tap switch was mounted on the front panel of the receiver. With this tap adjusted for loose coupling (only a few turns in the antenna coil), good reception of high-frequency stations was obtained. With tight coupling (many turns in the antenna coil), good

reception at low frequencies was secured.

► After the introduction of this first successful single-dial receiver, many others followed. Some omitted the first variable condenser and the variometer was ganged to the variable condensers in the manner shown in Fig. 14A. Some omitted the variometer entirely, using one section of the ganged variable condenser to tune the antenna circuit broadly, and using a trimmer or small variable condenser in parallel to permit accurate adjustment when receiving weak distant stations, as shown in Fig. 14B. At this stage in development, almost all receivers were tuned in exactly the same manner; strong stations were tuned in with the single tuning dial, while weak stations were first tuned in with this dial and then the antenna circuit was adjusted for maximum output by means of the antenna vernier trimmer.

A, B AND C BATTERY ELIMINATORS

All early radio receivers up to the single-dial control sets obtained power



Courtesy U. S. Dont. of Agriculture

An Atwater Kent single-dial battery set with horn loudspeaker. One of the very popular early models, widely sold in both urban and rural areas.

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from batteries of one type or another. Dry cells built into convenient blocks of fifteen or thirty cells, with all cells connected in series, were used to supply d.c. voltages to the plates of the tubes. The storage batteries used for filament power had to be removed for recharging at intervals of one to three weeks; many set owners bought their own battery chargers for this purpose. Later, trickle (low rate) chargers were surges of current produced by the "jerk" tube action from damaging the tube and also to reduce the resulting noise.

B and C battery eliminators met with instant popularity, and the public began calling for A battery eliminators as well. Engineers went to work again, and by using identical electrical principles succeeded in developing a lowvoltage, high-current rectifier with a



FIG. 15. An early B and C battery eliminator using the Raytheon BH tube as a full-wave rectifier. Paper filter condensers were used at C, while C_B were buffer condensers used to take out some of the noise interference developed in these tubes.

placed on the market for use in homes; these charged the battery at a low rate during periods when the radio receiver was not in use.

► The average radio receiver of the time required replacement of B batteries at intervals of two or six months, and replacement of C batteries every six to twelve months. This constant replacement of batteries was a decided nuisance, and the radio public in general began calling for the newly-developed battery eliminators.

In 1925 the Raytheon Company of Cambridge, Massachusetts, perfected the cold-cathode gaseous rectifier tube known as the <u>Raytheon</u> BH rectifier; this tube was quickly adopted by receiver manufacturers for use in B and C battery eliminators. A typical combination B and C battery eliminator using this tube is shown in schematic form in Fig. 15. You will note that it is simply a standard full-wave rectifier circuit except that buffer condensers $C_{\rm B}$ are connected across the two diode sections of the tube to prevent ripple filter which satisfactorily eliminated the filament batteries. Various rectifying devices were used in A battery eliminators; many units were built with a special type A Raytheon cold-cathode rectifier tube or with the Tungar rectifier tubes originally developed for battery chargers, but the copper-oxide rectifier unit and the electrolytic rectifier unit were most widely used for A battery eliminators because of their lower cost and longer life. A typical A battery eliminator circuit



FIG. 16. Circuit of an A battery eliminator using a copper-oxide rectifier unit. B is a stepdown transformer, C is a full-wave or bridge type copper-oxide rectifier unit, while D and E comprise a low-voltage, high-current filter. (Condensers D are 2000-mfd., 25-volt electrolytic condensers.) Rheostat F controls the output voltage, while A is the main power switch.

using a copper-oxide rectifier unit is shown in Fig. 16.

THE A.C. RECEIVER

A person who went into a radio shop late in 1927 or early in 1928 to purchase a radio receiver had to choose five different and separate items (in addition to an antenna) if he desired a complete installation:

1. A radio receiver in its table model cabinet.

2. A loudspeaker.

S. A set of tubes.

4. An ABC power pack.

5. A suitable table on which to place the receiver and loudspeaker, with a shelf underneath for the ABC pack.

Each of these five units was available in a number of different brands, and oftentimes several different models were put out by one manufacturer.

The next step in the evolution of radio receivers was the combining of the ABC eliminator with the receiver itself to give a complete a.c.-operated receiver. At first, the ABC eliminator pack was simply placed inside an enlarged receiver cabinet, which made a somewhat bulky receiver. The attention of design engineers was then directed toward improvement of the power pack.

► As far back as 1925, RCA Manufacturing Company had been building into their receivers special innovations which the rest of the radio manufacturing industry was unable, either because of patent rights or because of high cost, to use. One of these special features was the hot-cathode diode tube used as a half-wave rectifier in the power pack; this tube was capable of furnishing as much as 60 ma. of rectified direct current, making it possible to connect the filaments of the type UV199 tubes in series and feed them with power obtained from the regular B and C battery eliminators. RCA was also the first to operate filament type power output tubes directly from a low-voltage a.c. source in commercially built receivers.

► Many schemes for operating receivers from a.c. power lines were being tried at about this time. In 1928 Me-Cullough introduced the first a.c. tube. He reasoned that if-the electronemitting surface could be electrically separated from a source of heat, it would be possible to use alternating current for filament heating purposes without encountering trouble because of a.c. in the amplifying and detecting circuits. He designed the cathode as a metal sleeve coated with a good electron-emitting material, inside which was the heater filament, insulated from the sleeve by porcelain or some other insulating material. The McCullough tube, also known as the Kellogg tube (after its manufacturer) had its filament terminals at the top of the tube, with the other electrodes connected to standard tube base prongs.

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The McCullough tube worked well as an a.f. amplifier, an r.f. amplifier or as a detector. Shortly afterwards Mc-Cullough introduced a power amplifier tube. (The principle of separating cathodes and filaments to permit a.c. operation of tubes is still in use today, although, of course, many refinements in tube construction have been introduced since these first a.c. tubes.) For B and C eliminators, most set manufacturers used the popular BH coldcathode rectifier tubes.

▶ In 1928 RCA introduced their famous Radiola 17, an a.c.-operated receiver. At the same time they made available for general use three special a.c. tubes; the type 26 tube, having a filament which heated and cooled slowly and could, therefore, be operated from a low-voltage a.c. source when used in r.f. or a.f. amplifier stages; the type 27 heater type tube, having a separate cathode and designed primar-



FIG. 17. Early forms of the horn, cone and dynamic loudspeakers.

ily for detector stages; the type 80 fullwave hot-cathode rectifier tube. The types 12 and 71 power tubes, originally designed for battery receivers, were found suitable for a.c. operation in the power audio stage. The year 1928 marked the general introduction of a.c.-operated receivers.

DYNAMIC LOUDSPEAKERS

In the early days of radio, loudspeakers were made and sold independently of receivers, even after the introduction of a.c.-operated receivers. The early horn type loudspeaker, shown in Fig. 17A, used a single headphone as a driving unit; greatly improved units, with balanced magnetic driving unit and long exponential (curved) horns like that in Fig. 17B, soon appeared and met with widespread acceptance. Next came the cone type loudspeaker, a paper cone driven by a balanced magnetic unit at its apex or peak. The first cone loudspeaker was made up of two paper cones cemented together at their rims, with the driving unit inside, mounted at the apex of one cone and driving the apex of the other cone as shown in Fig. 17C.

Single-cone loudspeakers came next, with the edges of the cone attached to a fixed mounting ring by a soft leather washer. The driving unit was mounted inside, and was linked to the apex of the cone. Cone loudspeakers were made in various sizes ranging from 8 to 36 inches in diameter, with the large sizes naturally giving excellent bass response. Some were placed in metal housings similar to that shown in Fig. 17D.

▶ In 1925, Rice and Kellogg of the General Electric Company presented a technical paper describing their dynamic loudspeaker, a unit using a moving coil to drive a paper cone, the edges of which were clamped to a fixed rim; details of this loudspeaker appear in Fig. 17E. The edges of this cone were made of soft, flexible leather, permitting the entire cone to move. These men also brought out the importance of the baffle in securing good low-frequency response.

In 1925, RCA Manufacturing Company_used this first-dynamic loudspeaker with one of their superhetrodyne receivers, an a.c. set using the old UV199 tubes (with filaments in series, getting power from a diode rectifier) and two type 210 power tubes connected in push-pull for the output stage. Because of patent difficulties, however, other manufacturers were unable to take advantage of these new developments.

THE RADIO RECEIVER PATENT SITUATION

To describe things mildly, the radio manufacturing industry prior to 1929 was decidedly "in a mess." Largescale production of receivers was risky, for the important radio patents were in many different hands and there was always the possibility of lawsuits for infringement of patent rights. Rather than give away all their profits in the form of royalties to patent holders, many manufacturers took the risk of lawsuits, knowing full well that a patent showdown would come very soon.

The Radio Corporation of America, formed during World War I at the suggestion of the United States Government in order to centralize radio facilities, had grown by leaps and bounds after the birth of radio broadcasting. This organization acquired the basic patents of the old American Marconi Company, together with American Telephone and Telegraph, Westinghouse and General Electric radio patents, the patents of other companies which it absorbed from time to time, patents obtained through outright purchase or through exchange with other radio manufacturers, and patents resulting from research in RCA laboratories. This organization was thus in a position to produce radio receivers and equipment without fear of serious litigation; it controlled the basic regeneration, t.r.f., superheterodyne, a.c. power pack, dynamic loudspeaker, vacuum tube, and vacuum tube circuit patents.

Realizing that order had to come from the existing chaos if the radio industry was to prosper, RCA proceeded to license reputable manufacturers. These then could, by combining their own patents with those of RCA and other patent-holding companies to whom they were paying license fees, manufacture radio receivers with reasonable freedom from lawsuits.

Manufacturers thus joined together, so today all reputable receiver manufacturers, both large and small, operate under a common patent agreement.

Radio Comes of Age

With patent problems at last out of the way, the radio industry definitely entered the "big business" class. Receiving circuits had been developed up to the point where the demands of the public were pretty nearly satisfied for a time, and manufacturers now turned their attention to cabinets which would improve the appearance of their product.

Before 1929, a few radio manufacturers had recognized the desire of the public for a single cabinet which would house a receiver with all associated ap-



FIG. 18. Circuit of the Majestic Model 70B receiver, the hit of the 1929 radio season.

paratus. For some time, large furniture manufacturers had been producing adaptations of lowboy and highboy console cabinets with provisions for housing the radio chassis, the loudspeaker, and the ABC eliminator.

The production of complete radio receivers in several styles of console cabinets by the Grigsby-Grunow Company,* incorporating all of the desirable technical features then available, was the outstanding event of the 1929 radio season. The Majestic line met with instant popularity, and many of these early console types of cabinet receivers are still in use today. The circuit of this Majestic receiver, shown in Fig. 18, is of special interest in connection with our studies of how practical requirements changed receiver circuits. You will observe that there are three t.r.f. amplifier stages (with four tuned circuits), a grid-leak-condenser detector, an a.f. voltage amplifier stage, a push-pull audio output stage and a dynamic loudspeaker which received its field current from the power pack unit.

The Majestic receiver was entirely a.c.-operated, with type 26 tubes in each r.f. and a.f. voltage amplifier stage. It had a single tuning dial, with an additional control for tuning the antenna circuit exactly to resonance (by adjusting the inductance of the input coil) when receiving distant stations. A variable resistor which served as volume control coupled the antenna to the receiver input circuit. Each r.f. amplifier stage was neutralized. A very satisfactory sound output was obtained from the dynamic loudspeaker. An outstanding characteristic of this Majestic receiver was its excessive amplification of low notes; this false bass reproduction was pleasing to the public, and because of it, many radio listeners even today insist that the old Majestic was "king of them all."

▶ During the years of 1929, 1930, and 1931, the t.r.f. receiver with four or

^{*}Manufacturers of "Majestic" radio receivers.

even five tuned circuits reigned supreme. The public learned to rate the selectivity and sensitivity of a radio receiver in terms of the number of sections in the variable condenser gang. For a radio receiver to sell well during these years, it had to be a.c.-operated, have a dynamic loudspeaker, be housed in a suitable console cabinet, have single-dial operation, have an illuminated tuning dial, and have a push-pull audio output stage.

NEW TUBES

At about this time, radio engineers developed standards for selectivity, sensitivity, fidelity, and power output (with negligible distortion) for radio receivers. A host of new radio receiving tubes resulted from the many important technical advances which were made.

It was found that the type 26 tube gave entirely too much hum when used in the a.f. stages. The type 27 tube was improved so it could be used in amplifier stages, thus eliminating this trouble.

Greater and greater power output was being demanded by the public. Some manufacturers used the older type 10 high-voltage power amplifier tube, while others turned to the new type 50 output tube, but these highpower output tubes were in general reserved for the higher-priced receivers. Type 81 half-wave diode rectifiers used in either half-wave or full-wave rectifying circuits were generally found in the power packs of these receivers. To meet the public demand for low-priced receivers, manufacturers used the type 45 power output tubes in push-pull with about 250 volts on the plates. thereby securing as much power output as either a single 10 or 50 tube would give at 450 volts.

 \blacktriangleright The use of three and four stages of t.r.f. amplification, coupled with the



FIG. 19. Circuit of the Rediola 80, a nine-tube superheterodyne announced in 1931 by RCA Manufacturing Co., Inc. This same circuit was also used in the Graybar 700, General Electric 31 and Westinghouse WR5 receivers.

increased power of broadcast transmitters, made the sensitivity of the square law grid-leak-condenser detector unnecessary. Designers turned to the linear C bias detector to secure less distortion in the detector stage and increase the gain and selectivity of the last t.r.f. stage. About 1930, tube manufacturers introduced the type 24 screen grid tube, primarily for r.f. amplification purposes; this gave far more amplification than a triode and made the old neutrodyne circuit obsolete.

With increased r.f. gain in receivers. it became possible to couple the antenna loosely and inductively to the first tuned circuit and secure true single-dial operation without any antenna trimmers. The well-designed receiver of 1930 consequently contained three stages of t.r.f. using screen grid tubes, a linear type 27 or type 24 detector tube, and a push-pull type 45 output stage preceded in some cases by an a.f. voltage amplifier stage. By this time the dynamic loudspeaker was considered an essential part of any receiver. The power pack used a fullwave type 80 rectifier tube. Volume control was commonly obtained by varying the C bias on the r.f. amplifier tubes.

THE RISE OF THE SUPERHETERODYNE

The difficulties encountered in amplifying radio frequency signals above 200 kc. had long been recognized by radio engineers, particularly Armstrong, who had already contributed to the development of regenerative and super-regenerative receivers. The heterodyne method of reducing the frequency of the r.f. carrier practically suggested itself as a solution to this problem, and thus was born the superheterodyne principle of r.f. amplification, near the close of World War I. Armstrong obtained phenomenal gain with intermediate frequencies of 30 to 45 kc. His superheterodyne patents were acquired by RCA, and about 1923 they produced the Radiola Super Heterodyne receiver, in which a 45-kc. i.f. section employing iron-core transformers and a sharp filter was used. RCA claimed the superheterodyne circuit for its own, and continued to develop it to a high state of efficiency. A number of radio parts manufacturers produced i.f. transformers and special preselector and oscillator coils for home experimenters, and home-made superheterodynes thus were the only other source of these receivers.

Late in 1930, the superheterodyne patents were released by RCA for use among their patent licensees, and t.r.f. circuits faded out of the picture, slowly at first and then with a sudden sweep as the superheterodyne circuit climbed to its throne as king of receiver circuits.

▶ In 1931, the RCA Manufacturing Company brought out the Radiola 80, a receiver which set a standard of performance that even today is difficult to exceed. The circuit diagram for this famous receiver is given in Fig. 19. Note that the two i.f. amplifier stages had screen grid tubes. The oscillator and second detector used triode tubes, with the latter serving as a linear power detector and feeding two type 45 tubes in a push-pull audio power output amplifier circuit. The first detector was preceded by one stage of tuned r.f. amplification, with a double resonant circuit serving as antenna coupler, to prevent interference troubles. All of the features necessary for good selectivity, high sensitivity, good fidelity, interference suppression, and high power output, are present in this circuit; its development over a period of years was an indication that receiver designers were giving attention to what the public wanted.

MORE NEW TUBES

The inability of the C bias type of volume control to reduce the gain for strong signals without distorting them resulted (in 1932) in the introduction of the type 35 and type 51 variable mu tubes. These tubes had long E_{g} - I_{p} characteristic curves, so that very high negative C bias voltages were required to reduce the plate current to zero. Cross-modulation and modulation distortion effects were reduced by these tubes.

► The inability of screen grid tubes, even those of the variable mu type, to handle wide variations in plate voltage without secondary emission troubles practically forced the pentode r.f. amplifier tube into existence. In 1932 a new series of amplifying tubes, consisting of the type 56 triode, the type 57 pentode and the type 58 super-control pentode (which had a variable mu characteristic), appeared.

► The type 45 triode power amplifier tube did not have quite enough amplification to operate directly from a triode acting as linear C bias detector, unless special attention was given to the coupling transformer design. The type 47 power amplifier pentode came into existence to meet this deficiency; the 47 tube was used either singly or in push-pull circuits.

▶ When receivers were designed to be sensitive to weak signals, they invariably had too much gain for powerful local stations. For this reason the average broadcast listener had to tune with one hand on the tuning dial and the other on the volume control in order to prevent blasting when tuning from a weak to a strong signal.

To achieve more nearly ideal singledial receiver control, automatic volume control was introduced around 1932. The first a.v.c-controlled receivers contained an extra vacuum tube which acted as a C bias detector and produced the variable negative C bias voltage required for control of the r.f. amplifier tubes. The result was automatic reduction in gain when strong signals were tuned in, and a certain amount of compensation for fading signals.

► The acceptance of the diode detector for signal demodulation purposes made unnecessary the use of a separate a.v.c tube. The diode detector gave



Courtesy Stewart Warner A large table model set, really a full-sized set in a table-type cabinet.

considerably less distortion than the linear C bias detector, was able to handle higher r.f. voltages, and at the same time developed a negative d.c. bias voltage for automatic volume control purposes. Diode detectors are now practically universal in the second detector stages of superheterodynes. The lower a.f. output of the diode detector required the use of an audio voltage amplifier. Rather than use an extra tube, two diodes and a triode were built into a single envelope, known as the type 55 duo-diode triode; here one diode section served as second detector and the other supplied the a.v.c. voltage.

MIDGET, AUTO AND FARM RADIO RECEIVERS

Midget Receivers. The year 1929 was the beginning of a period of business depression. Radio, being a new industry and a home substitute for more costly entertainment, kept right on going through the first years of the depression. By 1932, however, the greatly decreased income of the buying public began to make inroads on the volume of sales of high-priced radio receivers. Manufacturers decided that low-priced receivers were their only salvation.

The first midget receiver to attract widespread attention was the Jackson-Bell table model receiver, introduced about 1931. This first midget receiver was a t.r.f. set using screen grid r.f. amplifier tubes followed by a screen grid detector and a type 45 power am-



Courtesy Adm

A modern type midget set.

plifier. The power pack used a type 80 full-wave rectifier tube.

Then came a flood of midget receivers, together with new tubes designed especially for them. In 1932, midget superheterodyne receivers were introduced. They invariably used the type 24 tube as a combination first oscillator-mixer-detector, so a single tube performed the entire frequency converter function.

Large manufacturers of radio sets began turning out both t.r.f. and superheterodyne table model receivers in various sizes ranging from the "overcoat pocket" and "cigar box" sets to



Courtesy Motorola

A true portable, carried by the strap which goes over the shoulder. Special small-sized batteries and tiny radio parts make such receivers possible.

large-size table model receivers, all complete including loudspeaker, and some even having built-in antennas. Midget receivers, limited in both space and cost, resulted in two innovations —the use of multi-function tubes (particularly pentagrid converter tubes), and universal a.c.-d.c. operation (eliminating the need for a power transformer and permitting use of the receiver in either d.c. or a.c. localities). Midget receivers are available now in a wide variety of styles, sizes, and even colors.

Auto Radios. Mobile (movable) radio equipment was by no means new, for it was being used extensively by the War and Navy Departments, as well as in airplanes. As far back as 1922, experimenters had been installing radio receivers in automobiles, but it was not until 1932 that radio manufacturers gave any considerable amount of attention to the auto radio receiver. <u>Mass</u> production technique was applied to the auto radio problem as another attempt to bridge the summertime drop in radio sales.

The first auto radios to appear on the general market included a radio chassis in a steel box, a remote tuning and volume control unit, either B batteries or a dynamotor for supplying the required high d.c. voltages, a separate loudspeaker, and suppressor resistors and condensers for the electrical system of the car to suppress ignition interference. The tube filaments (and the dynamotor) obtained power from the 6-volt auto storage battery.

It is interesting to note that auto radios went through the same development process as had home radios some years before. In each case the many separate units were gradually combined into one compact unit. The auto radio loudspeaker became an integral part of the radio chassis. The dynamotor was gradually replaced by the vibrator type of B supply unit, which also was built into the chassis. The superheterodyne became the standard auto radio circuit, and a.v.c. was introduced to compensate for fading. An entire auto radio set could now be bolted to the bulkhead or just behind the dash. Tuning controls which harmonized with the dash panel design of various car models were made available about 1935.

► Research work on auto radios naturally resulted in considerable benefit to radio receivers in general; perhaps the most outstanding contribution was the 6.3-volt line of vacuum tubes. The filaments in an auto radio receiver get their power from the 6-volt car storage battery, the voltage of which may vary from 5 to 7.5 volts under ordinary driving conditions. This wide fluctuation in voltage made it impractical to use the existing 5-volt or 2.5-volt filament tubes, for these required accurate control of filament voltage. The new series of 6.3-volt heater type tubes could be connected directly to the storage battery. They were built to withstand the strain of voltage variations and yet each drew only about .3 ampere of filament current.

▶ Builders of universal a.c.-d.c. receivers tried using these 6.3-volt tubes, with the filaments connected in series,



Courtesy Philco

A modern three-way receiver, portable type, which will operate from a.c. or d.c. power lines or from self-contained batteries. Modern, highly efficient battery type tubes make these sets possible.

and obtained quite satisfactory results. A new group of tubes, drawing the same filament current as the 6.3-volt line but using higher filament voltages, now made its appearance; this group included the type 43 power output tube (with a 25-volt filament), together with the 12Z3 (12.5-volt filament) and 25Z5 (25-volt filament) rectifier tubes, designed specifically for use in midget receivers where the filaments of all tubes were connected in series.

Farm Radios. Throughout the phenomenal rise of radio receivers, the attention of manufacturers had been focused on receivers designed primarily for either a.c. or d.c. power lines. Homes without electric power, particularly farm homes, were still struggling along with the old battery receivers like those shown in Figs. 8 and 9, purchasing new B batteries every few months and recharging storage batteries even oftener. To be sure, a few receivers were made with the low-drain type UV199 and UV120 tubes, but these tubes were entirely too fragile to meet with widespread acceptance.

► Late in 1930, the National Carbon Company introduced their air-cell battery, which could deliver slightly over 2 volts at a current drain of about .65 ampere for at least 1000 hours. When used as filament supply battery on a radio set operated an average of three hours a day, this battery had a useful life of about one year and required no recharging.

The air-cell battery was designed for use with the line of 2-volt tubes announced at about the same time by RCA. These tubes were rugged, small, and easily adapted to existing radio receiver circuits. It now became easy to design efficient battery receivers for farm homes, and the increased demand for these sets resulted in the production of a complete line of modern battery type tubes, far more efficient than early types.

Modern Receiver Types

After 1930, new tubes followed each other in rapid succession. In general these were produced either to satisfy a demand for tubes with special characteristics or to meet special power supply requirements.

With such a wide variety of tubes from which to choose, radio receiver designers now have little difficulty in designing a radio for any given power supply. It is perfectly possible to take a given signal circuit (r.f. amplifier, detector or audio amplifier) and adapt it to any power supply condition simply by rearranging the supply circuit and in some cases changing tubes. An example will show you how this is possible.*

► Consider the superheterodyne circuit shown in Fig. 20, which is a batterv-operated set. It uses a type 1A7 pentagrid converter tube which receives input power from the antenna through a transformer having a tuned secondary; this first tube acts as a combination oscillator and mixer-first detector. It is followed by an i.f. amplifier stage containing two i.f. transformers, after which comes a duplexdiode-triode tube functioning as a diode detector, a.v.c. tube and a.f. voltage amplifier. Then comes a pentode power audio amplifier stage having resistance - capacitance coupling. The supply connections for a battery-operated receiver are shown here, with all supply leads in heavier lines. The filaments of all tubes are connected together in parallel. All B+ leads can be traced to tube electrodes. then through the tubes to the filament circuit and to ground. All of the control grids in the r.f. section can be traced to the diode detector from which they receive a.v.c voltage. The power output tube grid can be traced to the

^{*}Experienced servicemen seldom attempt to change a receiver over from one type of power supply to another, for many problems are encountered. The various circuits are presented here only to illustrate the principles involved.



FIG. 20. A standard superheterodyne arranged to operate from batteries.

C— power supply lead. Either a magnetic or a permanent magnet type dynamic loudspeaker is used, so no d.c. power is required by the loudspeaker to produce a steady magnetic field.

▶ When this same signal circuit is adapted for a.c. operation, it will appear as shown in Fig. 21. Again the power supply circuit and all power supply leads are shown in heavier lines. Heater type tubes, designed to operate with a.c. filament voltages, have been substituted for the low-drain battery type tubes. The power pack, which can be designed for any a.c. line voltage and frequency, uses a type 5Y3 fullwave rectifier tube. The filaments of all tubes are connected together in parallel and are completely isolated from the signal circuits. When electrode voltages lower than the power pack voltage are required, resistors are used in the power supply leads to drop the voltage.



FIG. 21. The same basic circuit as in Fig. 20, arranged for operation from 110-volt a.c. power.

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All of the *control* grids in this circuit can be traced to ground through resistors either directly or through the a.v.c. tube circuit. Other grid electrodes and all plate electrodes can be traced to the point marked H.V., which is the highest completely filtered d.c. voltage in the receiver. An electromagnetic type dynamic loudspeaker is used, with the field coil acting as a choke in the power supply filter.

▶ When this same signal circuit is used for an a.c.-d.c. universal receiver, the circuit takes on the form shown in dentally grounded at a time when the chassis is connected to the "hot" side of the power line. No ground is used with this receiver. All other features of this universal receiver are similar to those for battery operation. ▶ The manner in which this basic receiver signal circuit is adapted for 6volt operation, such as for automobile and farm radio receivers, is shown in Fig. 23. The filaments of all tubes are here connected in parallel directly to the 6-volt storage battery. A synchronous vibrator, also connected



FIG. 22. The standard circuit of Fig. 20 arranged for universal operation from 110-volt a.c. or d.c. power lines.

Fig. 22. Here a diode rectifier tube is used, feeding into a filter which furnishes the d.c. electrode voltages. Supply leads are again shown in heavy lines; tube filaments are all connected together in series and are independently connected to the line voltage terminals.

Condenser C_A , in series with the antenna, allows r.f. current to flow but blocks d.c. and has a high reactance at power line frequencies. It prevents damaging the antenna coil primary in the event that the antenna is accito the battery, changes the d.c. to a.c., which is then stepped up by the power transformer and again converted back to pulsating d.c. by the vibrator. The pulsating d.c. is filtered and then fed to the various tube electrodes except the control grids. All control grids can be traced to ground through the a.v.c. circuit. The chassis is the common return for all supply circuits in this set.

Reading Circuit Diagrams. The circuits in Figs. 20, 21, 22, and 23 show that with minor signal circuit changes, a standard radio receiver can

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be adapted to any power supply. The analysis of these circuits has also made it clear that any circuit diagram can be traced or read in three distinct steps:

1. Trace the power pack and supply circuit leads_through the-signal circuits and tubes.

2. Trace the signal from the antenna to the loudspeaker (or image reconstructor in the case of a television receiver).

5. Trace any special control circuits, such as a.v.c. tone control, a.f.c., etc.

ALL-WAVE AND HIGH-FIDELITY RECEIVERS

Along with the expansion of radio broadcasting, other radio services on land, at sea and in the air grew to unusually large proportions. Police, aviation, commercial and amateur radio transmitters began crowding the frequency channels above the broadcast band frequencies; furthermore, broadcast stations in many foreign countries were sending out programs on these high frequencies, primarily to reach far distant colonies.

Radio enthusiasts soon learned that many interesting messages and programs could be picked up on the high frequencies (short waves.) About 1930, receiver manufacturers took recognition of the growing public interest in short-wave reception and began the production of short-wave converters. These converters consisted of a mixerfirst detector stage which, when connected to an ordinary t.r.f. receiver, used that receiver as the intermediate frequency amplifier, as second detector, a.f. amplifier and loudspeaker. A short-wave converter thus changed the broadcast band t.r.f. receiver into a short-wave superheterodyne receiver.

The addition of a separate unit to a receiver has always been followed by a combination of that unit with the receiver; when the superheterodyne circuit became firmly established as the basic receiver circuit, this combination . of broadcast and short-wave receiver units took place._With a superheterodyne, all that was required to make the change was a means of changing the preselector and oscillator coils for each frequency range desired. The i.f. value of all-wave receivers gradually rose from a low value of about 175 kc. to about 460 kc. in order to get away from the interfering signals which are inherent in a beat frequency system.



Uoursesy General Bleetrie A modern phono-redio combination receiver.

► The radio listening public has always been divided into two distinct groups, one containing those enthusiasts who prefer DX (long-distance) reception which is reliable and free from interference, and the other group prefering local reception only, but of the highest possible fidelity of reproduction. To be sure, the first group also desired a certain amount of fidelity, and consequently manufacturers had to make some compromise. While design engineers strove to improve selectivity and sensitivity, they attempted at the same time to bring up the fidel-



FIG. 23. The standard superheterodyne of Fig. 20 arranged to operate from a 6-volt storage battery.

ity as much as possible and eliminate interference troubles.

The demand for higher and higher fidelity continued, resulting in bandpass r.f. systems and flat-response a.f. amplifiers feeding into a high-fidelity loudspeaker with specially designed chambers or baffles. The problems of improving fidelity are by no means simple, for difficulties with voice, hum and volume range increase rapidly as fidelity is improved. Nevertheless, receivers are today on the market which will faithfully reproduce anything which can be put on the air by modern broadcast transmitters.

► Even though fidelity is all-important with certain listeners, they will not accept receivers which lack certain other desirable features. Therefore, the high-quality receiver of today incorporates a great many outstanding features, such as: high fidelity in the broadcast band; reception on all shortwave bands on which programs are regularly scheduled; compensated volume control; bass and treble control; automatic tuning of the pushbutton or telephone dial type, and tuning indicators.

TELEVISION

The radio industry, having gone through the code and sound signal development stage, now takes on the transmission of visual signals as well. The methods of modulating, broadcasting, and receiving visual signals do not differ in principle from the methods used for code and sound signals; only the ranges of modulation and carrier frequencies differ.

It is only natural that the well understood and thoroughly tested principles of the superheterodyne circuit should be extended for television service. Because of the high picture frequencies involved, ultra-high-frequency carriers are used. The preselector circuit in a television receiver must therefore tune to ultra-high-frequency carriers; the i.f. amplifier must handle a wide range of side frequencies; the detector circuit must have no frequency distortion, even at the highest picture frequencies, and the picture amplifier (if used) must likewise have no distortion. Special sweep circuits are needed to control the electron beam in the cathode ray image reconstructor tube.

Thus you can see how the amplitude-modulated radio receiver has passed through various stages of development, starting with the coherer, and moving in rapid succession to the basic circuits of crystal detectors, regenerative vacuum tube detectors, tuned radio frequency amplifiers, and finally to the superheterodyne circuit, which today is king of them all.

F.M. RECEIVERS

Frequency modulation broadcasting is the culmination of years of research and development work by Major E. H. Armstrong and other radio engineers seeking to provide a broadcasting service which is free from interfering noise, free from station interference, and has true high fidelity. These three basic listener requirements are now met more completely than ever before. With f.m., radio reception acquires a breath-taking naturalness, and audio frequencies as high as 15,000 cycles are reproduced faithfully.

F.M. receivers can be divided into three groups: 1, those designed only for f.m. reception; 2, those providing both a.m. and f.m. reception; 3, f.m. converters, which connect to the a.f. input terminals of an a.m. receiver.

Some sections of f.m. receivers employ entirely new circuits, while others simply use basic circuits which are suitably designed to handle all the frequencies encountered in an f.m. system. The a.f. sections and loudspeakers, however, can be the same in both f.m. and a.m. receivers because they handle the same types of signals. An f.m. receiver will usually be designed to give high fidelity, so as to take full advantage of the higher-quality signals put on the air by f.m. transmitters.

Looking Ahead. Modern receivers of the superheterodyne type, television and f.m. receivers are just mentioned here. You are going to study them in detail in later lessons. Also, receiver refinements, such as a.v.c., tone controls, a.f.c., push-button systems, etc., are all to be thoroughly covered in future lessons.

THE N. R. I, COURSE PREPARES YOU TO BECOME A
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Lesson Questions

Be sure to number your Answer Sheet 18FR-2.

Place your Student Number on every Answer Sheet.

Send in your set of answers for this lesson immediately after you finish them, as instructed in the Study Schedule. This will give you the greatest possible benefit from our speedy personal grading service.

- 1. What seven important technical requirements, demanded by the public, control the circuit chosen for any particular receiver?
- 2. Who predicted mathematically that an oscillating current will radiate electromagnetic waves which are identical with light waves except for frequency?
- 3. Why were loading coils sometimes inserted in the tuning circuits of early receivers?
- 4. Why is the super-regenerative receiver unsatisfactory for use in congested bands?
- 5. Why were grid suppressors used in some of the earlier t.r.f. receivers?
- 6. What objectionable effect compelled set designers to drop the idea of untuned antenna systems in single-dial receivers?
- 7. In what radio apparatus was the Raytheon BH rectifier first used?
- 8. In addition to an antenna, what five separate items did a person have to select in the early part of 1928 in order to have a complete radio receiver installation?
- 9. In what year were a.c.-operated receivers generally introduced?
- 10. Name the three distinct steps involved in tracing or reading a circuit diagram.

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THE FABLE OF DISCOURAGEMENT

Once upon a time, so the story goes, the devil held a sale. To any one who would pay the price, he offered the tools of his trade. On a table, each with its price label, were *hatred*, *despair*, *sickness*, *jealousy*, *greed*, and all the other causes of unhappiness.

Off to one side, however, lay a harmless-looking wedge-shaped instrument marked discouragement. It was old and worn, but priced the highest of them all. When asked why the price was so high, the devil replied: "Because this tool is one I can use so easily. Nobody knows that it belongs to me, so with it I can open doors that are immune to all other tools. And once inside, I can finish the job with almost any of the other tools!"

Few people know how small is the margin between failure and success. Frequently, the separation is just the width of that one word—discouragement.

You can combat discouragement by cultivating confidence in yourself. Whatever you may desire of life—whatever your goal may be—you have only to work for it wholeheartedly, confidently, with that one goal always in mind, and you will reach it.

J. E. SMITH

MANUAL AND AUTOMATIC VOLUME CONTROLS

aland W. Carup

19FR-2

NATIONAL RADIO INSTITUTE ESTABLISHED 1914 WASHINGTON, D. C.



STUDY SCHEDULE No. 19

For each study step, read the assigned pages first at your usual speed. Reread slowly one or more times. Finish with one quick reading to fix the important facts firmly in your mind, then answer the Lesson Questions for that step. Study each other step in this same way.

The purpose of a volume control in a receiver, and the difference between automatic and manual volume controls, is explained here. Then, there is a discussion of R.F. and A.F. voltage control systems. Answer Lesson Questions 1, 2, and 3.

Г] 2.	R.F.	Gain	Controls	Pages	6-1	l
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When automatic volume control systems are not used, an R.F. control is necessary. There are many types of gain controls—some vary the load while others change operating voltages. Of these, the C bias and the antenna-C bias controls are the most popular. Answer Lesson Question 4.

3. Automatic Volume Controls.... Pages 11-15

This is an explanation of the purpose and function of A.V.C. systems, how the A.V.C. voltage is filtered, and the importance of the time constant values of the filter. Answer Lesson Questions 5, 6, and 7.

Here are descriptions of typical A.V.C. circuits. There are many types; some use separate tubes, and some have tapped voltage dividers. Answer Lesson Question 8.

Few volume controls are made so that the resistance variation is linear. Most of them have either a left-hand or a right-hand taper. The proper taper is very important—sometimes more so than the resistance value itself. Study this practical section with great care. Answer Lesson Questions 9 and 10.

6. Mail Your Answers for this Lesson to N.R.I. for Grading.

7. Start Studying the Next Lesson.

MANUAL AND AUTOMATIC VOLUME CONTROLS

Why Volume Controls Are Needed

THE most important control in a radio receiver is unquestionably that which allows the listener to choose one desired station from the many which may be on the air at any one time. Next in importance is the volume control, which directly or indirectly controls the acoustical level or volume of the loudspeaker sound output. Volume controls are of two general types: 1, automatic volume controls; 2, manual volume controls. All radio receivers require manual volume controls, even when equipped with automatic volume control.

Automatic Volume Control. Automatic volume control (commonly abbreviated A.V.C.) keeps the output volume level of a radio receiver essentially constant despite variations in R.F. signal input voltage. These variations may occur while the radio receiver is tuned to a single station (fading) or while the receiver is being tuned from station to station.

The R.F. input signal voltages which modern receivers are called upon to handle may vary from 1 microvolt for low-power or distant stations to 1 volt for high-power local stations, and thus the strongest signal handled is a million times greater than the weakest signal. If the gain of the R.F. amplifier were *fixed* at a value sufficient for reproduction of the weakest signals, there would be blasting (excessive volume) when the set was tuned to a strong signal. A.V.C. counteracts this blasting by raising the gain of the R.F. amplifier for weak signals, and lowering the gain automatically and almost instantly for strong signals.

Blasting is particularly annoying

when the listener is tuning rapidly from station to station while searching for a desired program. Furthermore, blasting may be accompanied by overloading of the loudspeaker and of one or more stages in the receiver, with resulting distortion. A.V.C. prevents this overloading by reducing the gain of the R.F. amplifier for strong signals.

Programs received from distant and semi-distant stations, where the radio waves are totally or partially reflected from the sky before reaching the receiving antenna, are subject to fading. In other words, the strength of the signal at the receiving location varies considerably from minute to minute. A.V.C. can compensate for the effects of fading by raising and lowering the gain of the R.F. amplifier just enough to maintain the desired acoustical output, provided the signal does not become excessively weak. Automatic volume control is thus desirable in a radio receiver for these three important reasons: 1, it prevents blasting; 2, it prevents overloading; 3, it minimizes fading.

Manual Volume Control. A manual volume control is essential in a radio receiver for a number of reasons. Low volume may be desired when a musical program is used as atmospheric background while eating, reading, playing games, or carrying on conversation. Increased volume may be desired for news broadcasts, talks by important persons, radio plays, popular all-star broadcasts, symphonic music, or music for dancing; furthermore, some people can hear better than others, and the volume must be adjusted accordingly.

But what, exactly, is a volume con-

trol? To the average non-technical person, any control which permits him to change the volume of the sound coming out of the loudspeaker is a volume control. Radio men know, rent swing over non-linear regions of the E_{g} - I_{p} characteristic of the tube. Because of this distortion, manual volume controls in receivers not having A.V.C. must be R.F. gain controls;



Fig. 1. Typical manual volume control with accessories (e unit mede by the Yexley Menufacturing Division of P. R. Mallory & Co., Inc.). A feature of this control is the adjusteble grid bias plate (1) which permits adjusting the minimum resistance to 100, 200, 300, 400 or 500 ohms for fixed C bias purposes. Numbers on housing and indicating mark on plate tell which setting is used. With this plate removed, the control can be turned to zero resistance. The 3" shaft can be cut to any desired length with a file or hacksaw, or can be lengthened when necessary by using an extension shaft (6). A small metal spacer (3) is supplied with Yaxley controls for use with knobs designed for shafts having only 1/32" deep flats. The back cover plate of the control housing is easily removed to permit attachment of a power switch (4). The special nut (7) is used for mounting this nut or the regular flat nut shown on the shaft, as pliers may wear off the corners of these nuts or scratch the panel.

however, that volume controls are either voltage controls or gain (amplification or sensitivity) controls. In receivers which have automatic volume control (commonly abbreviated as A.V.C.), the manual volume control is always an A.F. voltage control; in all other receivers the manual volume control is always an R.F. circuit control. (Regardless of whether the control is an R.F. or an A.F. type, the physical appearance will be almost the same. A typical control is shown in Fig. 1.)

There is a definite limit to the amount of signal voltage which can be handled by each stage in a radio receiver. Excessively high input voltage to a stage results in distortion, because high input signals make the plate curthey can then be adjusted to prevent overloading of any receiver stage.

TYPES OF MANUAL VOLUME CONTROLS

The manual volume controls found in radio receivers are of three types, which may be used either singly or in combination: 1, R.F. voltage controls, which reduce the voltage of the R.F. input signal either before it reaches the first R.F. amplifier stage or before it enters an R.F. stage which may overload; 2, A.F. voltage controls, which reduce the A.F. voltage output of the demodulator stage or reduce the A.F. voltage at some point in the A.F. amplifier; 3, R.F. gain controls, which reduce the gain (amplification) of an **R. F.** amplifier stage either by varying the load on that stage or by varying the mutual conductance of the tube in that stage. We will now consider the advantages and disadvantages of each of these forms of manual volume controls as applied to practical radio receiving circuits.

R.F. VOLTAGE CONTROLS

R.F. voltage controls can either vary the R.F. voltage which is fed from the antenna to the input of the R.F. amplifier or can vary the R.F. voltage which is fed from one R.F. movable contact arm connected to a third terminal (labeled C); the significance^{*} of these R, L and C (right, left and center) designations for these volume control potentiometers will be explained to you later. The value of the voltage fed to the following R.F. circuit depends upon the position of the movable arm of the potentiometer.

With the circuit of Fig. 2A, the fixed terminals of the potentiometer were connected to antenna and ground, and the variable tap on the potentiometer was connected directly to the grid of the first R.F. amplifier tube. The potentiometer thus con-



Fig. 2. Typical manual volume control arrangements which vary the amount of R.F. voltage transferred from one circuit to another. The resistance values indicated for V.C. are maximum values; the minimum resistance in each case is zero. The designations R, L, C, R.H. and L.H. are explained later in this text-book.

amplifier stage to another. These controls were widely used in older receiver circuits for the purpose of controlling volume, the antenna circuit control having been the more popular.

Five typical R.F. voltage control circuits are shown in Fig. 2. As you can see, the control device in each case was a potentiometer with two fixed terminals (labeled R and L), and a

trolled the R.F. voltage which was applied to the grid of this tube. This circuit had no selective or signalboosting properties; therefore, only the characteristics of the antenna determined which signals would be favored. The antennas used at the time were generally of such a length that they were in resonance at the higher frequencies in the broadcast band, and consequently these higher frequencies were favored.

To make the simple antenna circuit volume control of Fig. 2A respond equally well to low and high frequencies, inductance L_{A} was connected between grid and cathode of the first R.F. tube in the manner shown in Fig. 2B. The inductance and distributed capacity of coil L_{A} acted together to form a parallel resonant circuit whose resonant frequency was in the lower region of the broadcast band. This resonant circuit acted as a high impedance to low frequency signals, developing maximum R.F. voltage for transfer to the grid of the tube, and acted as a capacitive reactance path to ground (a shunting capacity) for higher frequency signals. Bv choosing the proper ratio of inductance to capacity for coil L_{A} , it was possible to make the antenna circuit respond uniformly to both low and high frequencies in the broadcast range.

The volume controls shown in Figs. 2A and 2B worked well in receivers located at a distance from powerful broadcast stations, but in the vicinity of a powerful station the capacity path between the leads and parts of the volume control was sufficient at radio frequencies to transfer the signal of the local station to the grid of the first tube at all volume control settings. When the volume control was turned up to receive a weak signal, the signal of the local station overloaded the first R.F. amplifier tube, making this tube act as a detector which modulated the weak signal with the program of the strong signal; cross-modulation resulted, with both the desired and the undesired programs being reproduced by the loudspeaker.

The next move of the radio receiver designer was to provide selectivity for the antenna circuit in order to counteract cross-modulation effects. One of these selective circuits is shown in Fig. 2C. As you can see, the voltage output of the volume control was fed into the primary of an R.F. transformer, and the resulting R.F. current induced a stronger R.F. voltage in the tuned secondary of this transformer for transfer to the grid of the first tube. The chief drawback of this circuit was the fact that whatever resistance existed between terminals Cand L of the potentiometer was reflected into the resonant circuit as a resistance; this made selectivity very poor at low volume control settings.

The modified potentiometer connection shown in Fig. 2D gave improved selectivity. Note that the variable tap of the potentiometer was grounded and that the entire potentiometer resistance was connected across the primary of the R.F. transformer. This resulted in a constant and high resistance being reflected into the resonant circuit, keeping the selectivity essentially constant. Antenna and ground connections were made to points L and C on the potentiometer. As the resistance between these two points was increased, more of the antenna current flowed through coil L_{A} and therefore a higher R.F. voltage was induced in the resonant circuit.

The usual form of R.F. voltage control used between two R.F. stages was that shown in Fig. 2E. Here the voltage developed across series resonant circuit L_1 - C_1 was fed to the two fixed terminals of the volume control potentiometer, and the variable tap of the potentiometer was connected directly to the grid of the following R.F. amplifier tube. The potentiometer resistance was generally between .25 megohm and 1 megohm, so the load placed upon the resonant circuit by the volume control was negligible. One disadvantage of this circuit was the fact that body capacity, such as placing the hand near the volume control, tended to detune the circuit; this made it necessary to use a long shaft of insulating material between the potentiometer and its control knob. Furthermore, this volume control and the antenna circuit R.F. voltage con(acting as load).

When an audio transformer is used at the input, a common form of A.F. voltage control is that shown in Fig. SA. Here audio transformer T couples the A.F. signal source to volume control potentiometer V.C. In any



Fig. 3. Typical manual volume control arrangements which vary the amount of A.F. voltage transferred from a source of A.F. voltage to a following circuit.

trols were quite noisy, for they controlled low R.F. voltages in high-gain circuits.

A.F. VOLTAGE CONTROLS

Several different ways of connecting manual volume controls into audio frequency amplifier circuits are shown in Fig. 3. Controls such as these are essential in the audio amplifiers of public address systems, in intercommunicating systems, in electric phonographs and in all radio receivers which employ A.V.C. in the R.F. amplifier. Basically, an A.F. voltage control varies the amount of audio frequency voltage which is fed from one circuit (acting as source) to another circuit where the signal source has a high impedance, approaching that of the volume control, the transformer may be omitted and the source connected directly to terminals R and Lof the potentiometer. Any desired portion of the voltage applied across the potentiometer is fed to the grid and cathode terminals of the following A.F. amplifier tube.

With resistance-capacitance coupled audio amplifiers, the arrangement of the A.F. voltage control is as shown in Fig. 3B.

When inserting a volume control in the input of a circuit which is acting as a load for an A.F. signal source, the volume control must not alter the impedance match between the source

and load. In cases like this, special constant-impedance volume controls or pads are used; these consist of two or more variable resistors which are adjusted simultaneously. The circuit in Fig. 3C is an example; here a socalled-T-pad volume control having three variable resistors (this name results from the fact that the resistors are arranged to form the letter "T") is so connected that the resistance across the input terminals and the output terminals of the T-pad remains constant (at 200 ohms for a 200-ohm Tpad or at 500 ohms for a 500-ohm Tpad) regardless of the setting of the control knob. Only the A.F. voltage across the output terminals of the Tpad varies with the control knob setting. This subject is covered in greater detail in advanced lessons; for the present it is sufficient for you to know what this type of A.F. voltage control is and why it is used.

A type of A.F. voltage control which you will constantly encounter in radio receivers using diode detectors and A.V.C. is shown in Fig. 3D. Here the volume control serves as load for the diode demodulator or second detector. The R.F. signal is by-passed around this volume control by condenser C_{p} ; this condenser also serves to smooth out the rectified R.F. voltage, producing the desired A.F. signal across the volume control. The A.F. voltage between points L and C is fed to the grid of the first A.F. amplifier tube (in the same envelope as the diode detector) through D.C. blocking condenser $C_{\mathbf{K}}$. Other types of A.V.C. circuits in which manual volume controls are incorporated will be considered later in this book.

R.F. Gain Controls

Volume Controls Which Vary the Load in an R.F. Circuit. In the earliest battery-operated receivers, volume was controlled simply by varying the filament current of one or more tubes in the receiver. With the advent of A.C.-operated receivers, volume control became a more difficult problem, for varying the filament currents in heater type tubes did not give a satisfactory control over volume.

One of the first manual volume control schemes used in A.C. receivers was that shown in Fig. 4, which involved placing a variable resistor across the plate load resistance of an R.F. stage. Without this variable resistor the tuned secondary R.F. transformer made up of L_1 , L_2 and C_2 offered a definite load in the plate circuit of tube VT_1 , and the stage consequently had a definite amplification or gain. Shunting a volume control across L_1 permitted reduction of the plate load impedance and the gain to any desired lower value. Because of the fact that decreasing the load impedance increased the curvature of the $E_{\rm g}$ - $I_{\rm p}$ curve, making strong R.F. signals swing over non-linear portions of the curve, this scheme could be used only in the plate circuit of an R.F. stage which handled low R.F. voltages. This method of controlling volume also served as a control on regeneration, for it varied the A.C. plate voltage and thus varied the feed-back voltage.

Volume Controls Which Vary the Screen Grid Voltage in an R.F. Stage. The amplification of a vacuum tube amplifier can be controlled to a considerable extent by varying the D.C. voltages applied to the electrodes of the vacuum tube used in the stage. Any change in electrode operating voltages which affects the mutual conductance of the tube will also affect the amplification provided by the stage. This is true because, as you will recall, mutual conductance is equal to A.C. plate current divided by A.C. grid voltage. Reducing the A.C. plate current therefore reduces the mutual



Fig. 4. This volume control varies the impedance of the load in an R.F. circuit and thus varies the gain of the R.F. amplifier.

conductance and, at the same time, reduces the A.C. voltage which is developed across the plate load for transfer to the following stage.

As a general rule, reducing the D.C. plate voltage, reducing the D.C. screen grid voltage or increasing the negative C bias voltage will reduce the mutual conductance of a tube. Making the suppressor grid negative with respect to the cathode and increasing its negative voltage will also decrease the mutual conductance. It is not customary, however, to vary the suppressor grid voltage for the reason that very large changes are required in voltage in order to secure any appreciable reduction in mutual conductance.

The usual method for controlling the screen grid voltage was that shown in Fig. δA ; the volume control potentiometer was made part of a voltage divider connected across the plate supply source, with resistor $R_{\rm L}$ serving to determine the maximum screen grid voltage. Condenser $C_{\rm s}$, the screen bypass condenser, also helped to prevent noisy volume control action, for it shunted the L-C section of V.C. and thus by-passed to ground any current impulses created by a varying contact resistance at C.

Volume Controls Which Vary the Plate Voltage of an R.F. Amplifier Tube. A method used in early battery receivers for varying the plate voltage of a tube as a means of controlling volume was that shown in Fig. δB , where a variable resistor was placed in series with the plate supply voltage and the R.F. plate load. This method was advantageous in that it did not place any unnecessary drain upon the B batteries, but had the disadvantage that a very high-resistance rheostat, which burned out easily, was required.



Fig. 5. Typical manual volume control arrangements which vary the mutual conductance of an R.F. amplifier tube by varying the screen grid voltage (A) or the plate voltage (B and C).

Plate voltage can also be controlled by using a potentiometer connected across the plate voltage supply in the manner shown in Fig. 5C. With this potentiometer across the B battery at all times, there was naturally a higher current drain than in the case of Fig. 5B, but the potentiometer was lower in resistance, making possible a more rugged design and consequently reducing the tendency to burn out.

Unfortunately, these methods of controlling volume by varying screen grid and plate voltage did not prove entirely satisfactory. Reducing these electrode voltages any reasonable amount affected the dynamic E_{g} - I_{p} characteristic of the tube enough to cause distortion at low volume control settings. These circuits were quickly replaced by the still popular C bias method of volume control, which will now be considered.

Volume Controls Which Vary the C Bias Voltage of an R.F. Amplifier Tube. The most effective way of varying the mutual conductance of a tube in order to control volume is byvarying the control grid voltage (the C bias voltage). This method of controlling volume proved so satisfactory and came into such widespread use that tube designers developed special remote cut-off tubes, first the variable mu screen grid tube and later the super-control pentode tube, which gave even better results. These tubes allowed the A.C. grid voltage to swing over an essentially linear portion of the dynamic $E_{\rm g}$ - $I_{\rm p}$ characteristic curve at all volume control settings. A number of typical C bias volume control circuits are shown in Fig. 6; although triode tubes are indicated for simplicity, the circuits also apply to screen grid and pentode tubes.

A scheme widely used in early battery receivers for varying the C bias voltage was that shown in Fig. 6A, where the negative C bias voltage was increased and the filament current was reduced simultaneously (by increasing the resistance of volume control V.C.) in order to reduce the mutual conductance of the tube. When the filaments of several tubes are in parallel and the grid return lead for each tube circuit is grounded, a volume control such as this will affect all such parallel-fed circuits.

Another C bias voltage control for battery receivers, used when filament voltage was fixed in value, is that shown in Fig. 6B. Here potentiometer V.C. and resistor R_1 were placed in series across a C battery which was connected with the indicated polarity. The value of R_1 must be low enough to develop the minimum C bias required for the circuit. (A high-resistance load on the C battery will make it last longer than the B batteries; with normal C bias and low B voltages, certain tubes will distort severely. For this reason, values for V.C. and R_1 were chosen which were low enough to make the B and C batteries run down at approximately the same rate.) Switch SW opened both the A and C battery supply circuits in order to prevent current drain when the receiver was not in use.

The basic C bias voltage volume control for heater type tubes is that shown in Fig. 6C. The volume control is placed in series with the usual cathode bias resistor $R_{\rm c}$, with condenser $C_{\rm c}$ connected between the cathode and the movable tap of the volume control. Increasing the resistance setting of the volume control increases the C bias voltage and therefore reduces volume; resistor R_c determines the minimum C bias voltage. The chief disadvantage of this circuit is that as the negative C bias voltage is increased, the plate current which flows through the cathode resistor combination is greatly reduced, counteracting the effect of the volume control. For example, if a cathode bias resistance of 600 ohms gives a 3 volt negative bias, it may be necessary to increase this resistance to as much as 25,000 ohms in order to secure three times as much bias voltage (-9 volts).

Where the movable tap of the volume control is grounded, as in Fig. 6C, receiver manufacturers will often use a potentiometer in which the movable tap or middle terminal C is grounded internally to the metal shaft of the made essentially independent of the plate current. The volume control now has greater control over the C bias voltage and can be considerably lower in value than in the case of Fig. 6C.

In some receivers the volume control is actually a part of the power pack voltage divider system or is an additional voltage divider across a part of the main voltage supply in the manner shown in Fig. 6E. Here volume



Fig. 6. Typical manual volume control arrangements which vary the mutual conductance of an R.F. amplifier tube (thereby varying the gain of the R.F. amplifier) by varying the C bias voltage applied to the control grids of one or more R.F. tubes.

control. This eliminates the need for making a soldered connection to the middle terminal, for mounting the potentiometer on the metal chassis automatically grounds the housing.

An improved C bias volume control arrangement is shown in Fig. 6D. By having between the plate voltage lead and cathode a bleeder resistor R_B which forces or "bleeds" a current through the C bias resistor R_c at all times, the volume control current is control potentiometer V.C. varies the voltage between the cathode and the B— terminal, with resistor R_1 determining the minimum C bias voltage and resistor R_2 determining the maximum C bias voltage. The total resistance of R_1 , V.C. and R_2 in series across the power pack determines the value of bleeder current which flows at all times. Both the plate current of the tube and the bleeder current flow through R_1 and the R-C section of the
volume control, producing a C bias voltage which is dependent upon the position of movable tap C.

4

Volume Controls Which Vary Both the Antenna Input and C Bias Voltages. We cannot pass over C bias



5000 TO 200000 (R.H.)

Fig. 7. A manual volume control arrangement which varies both the antenna input R.F. voltage and the C bias voltage of the first R.F. stage in order to change the R.F. amplifier gain.

voltage volume controls without pointing out that with the earlier triode and screen grid tubes, increasing the negative C bias beyond a certain point changed an R.F. amplifier stage into a two types of volume control circuits in solving this problem.

One typical combination volume control circuit is shown in Fig. 7. Here fixed resistor R_c controls the minimum C bias voltage and hence the maximum amplification of the stage. That section of the volume control potentiometer between points C and R also serves as cathode bias resistor, while the ternaining section shunts the input transformer primary winding. Moving contact C away from point R increases the C bias voltage and at the same time reduces the resistance across the primary winding, thus giving a double reduction in R.F. signal gain.

Variations of Manual Volume Control Circuits. You are now familiar with the important basic manual volume control circuits. Of course you will encounter many variations of these circuits in radio receivers, but as a rule these variations will be easily recognized as one of the types which you have studied.



Today it has become an almost universal practice among receiver designers fo make the manual volume control or some other manual control serve also as an on-off switch. Earlier forms of dual potentiometer-switch arrangements appear at the left (a Carter unit) and in the center (a Yaxley unit); in both of these, rotation of the control knob to the extreme counter-clockwise position caused an arm on the control shaft to flip the toggle switch to its off position. At the right is a modern universal replacement volume control (a Philco unit) with a special power switch attached to its cover plate for this same purpose.

detector, with resulting modulation distortion and cross-modulation on strong signals. With variable mu screen grid tubes and super-control pentode tubes this is naturally no longer a problem, but it will be of interest to see just how receiver manufacturers combined Before beginning our study of automatic volume control circuits, we want to state definitely that with A.V.C.controlled receivers, the manual volume control must be *in the audio system*, except in those cases where the manual control is of a type which changes the effectiveness of the A.V.C. control. You can readily understand that any attempt to reduce the R.F. output of the R.F. amplifier manually

would simply cause the A.V.C. system to increase the amplification and counteract the change in the manual volume control setting.

Automatic Volume Controls

A.V.C. action is basically simple. When the carrier level of the R.F. input signal is excessively high, A.V.C. lowers the gain of the R.F. amplifier by increasing the negative C bias voltage on one or more R.F. amplifier tubes. This is accomplished automatically by rectifying the R.F. carrier signal, then filtering out all but the resultant D.C. voltage, whose value is always proportional to the R.F. carrier level. This D.C. voltage is so applied to the grids of the R.F. amplifier tubes that increases in carrier level make the grids more negative. The R.F. amplifier gain is thus reduced enough to keep the R.F. amplifier output level essentially constant and prevent overloading of any tubes. Likewise, reductions in input carrier level result in less negative C bias voltages and greater amplification.

It is clearly impossible for A.V.C. to maintain the R.F. carrier level perfectly constant at the output of the R.F. amplifier, for it is the change in this level which produces the change. in negative C bias voltage required for automatic volume control. With proper design, however, A.V.C. can keep the carrier level constant enough for all practical purposes. Exact control of carrier level is not required, for the output voltage of a receiver can be increased or decreased about 40 per cent before the change can even be detected by the human ear. When working with A.V.C.-controlled circuits, then, never depend upon your ears as a judge of performance; always use an output meter or some other type of indicator.

The action of a receiver having A.V.C. is best represented by overload curves like those in Fig. 8, which are obtained by plotting the R.F. input voltage of the receiver against the A.F. output voltage of the demodulator or second detector. Curve 1 in Fig. 8 represents the overload characteristics of a receiver not having A.V.C. Notice that overloading takes place at an R.F. input voltage of about 100 microvolts



Fig. 8. Overload characteristic curves for three different superheterodyne radio receivers, obtained by feeding various R.F. input voltages into each receiver (the manual volume control being set for maximum volume) and measuring the A.F. output voltage of the second detector. Curve 1 is for a sensitive receiver which does not have A.V.C. Curve 2 is for the same receiver with ordinary A.V.C. added. Curve 3 is for the same receiver with delayed A.V.C. added. These curves illustrate receiver performance in general, and will naturally vary greatly with different receivers.

in this particular example. This is shown by the output falling off when the input is increased further. When this receiver is equipped with A.V.C., its overload characteristic is represented by curve 2. You can readily see that with A.V.C., the receiver will handle all carrier signal levels below 100,-000 microvolts (.1 volt) without overloading. On the other hand, however, this A.V.C. curve shows that the sensitivity of the receiver will be considerably lower with A.V.C. than without it __ for medium-strength signals. (R.F. input voltages in the region between about 10 and 100 microvolts do not deliver as much output.) Let us see how this drawback of A.V.C. is overcome in some receivers.

Delayed A.V.C. Since ordinary A.V.C. circuits prevent the maximum amplification of the receiver from being effective at low carrier levels, and since A.V.C. action is not particularly required at these low carrier levels since they cannot possibly cause overloading, the receiver designer simply arranges the A.V.C. system so it is inactive until the carrier level reaches a definite and fairly high value. This arrangement, which utilizes the full sensitivity of the receiver at low carrier levels, is known as delayed A.V.C. The R.F. input voltage level at which A.V.C. action begins_is called the threshold point or the threshold voltage. The overload characteristic curve for a receiver having delayed A.V.C. is represented by curve S in Fig. 8. Notice that it follows curve 1, that for a receiver without A.V.C., up to the threshold voltage, after which it levels out and effectively prevents overloading.

Since the A.F. output voltage of the second detector in a receiver having delayed A.V.C. is quite high at the threshold point, further increases in output voltage should be prevented as much as possible. To do this, the negative C bias voltage produced by A.V.C. action should be applied to as many R.F. tubes as possible. If even this is insufficient, the A.V.C. voltage can be amplified by an extra vacuum tube stage, giving what is known as an

amplified delayed A.V.C. circuit.

Definitions. Those R.F. amplifier stages in an A.V.C. receiver which are to vary in gain as the incoming carrier level varies are called the A.V.C.-controlled stages or simply the controlled stages. The tubes in these controlled stages are called A.V.C.-controlled tubes or simply controlled tubes. The vacuum tube or tube section which converts the amplified modulated R.F. carrier into a D.C. voltage suitable for A.V.C. purposes is called the A.V.C. tube or A.V.C. stage, and the D.C. voltage is called the A.V.C. voltage. The A.V.C. voltage may be produced as a part of the action of demodulation, or may be produced independently by an extra stage in the receiver.

SIMPLE DIODE DETECTOR-A.V.C. CIRCUITS

A simple diode demodulator or detector circuit like that shown in Fig. 9A is not only capable of separating the modulation signal from the R.F. carrier, but can also produce the negative C bias voltage required for A.V.C. purposes. The modulated R.F. carrier signal at points 1 and 2 in the final I.F. amplifier stage passes through the final resonant circuits, L_p - C_p and L_s - C_s , and is applied directly to the plate and cathode of diode detector tube VT_2 . Condenser C_D offers no opposition to this signal, for it has a low reactance at radio frequencies.

The modulated R.F. carrier is rectified by the diode tube, since this tube allows current to pass only in one direction; the wave form of the current passing through this tube is therefore like that shown in Fig. 9B. The charging and discharging action of condenser $C_{\rm D}$ on this pulsating current passing through $R_{\rm D}$ serves to filter out the R.F. variations, making the voltage across $R_{\rm D}$ have the wave form shown in Fig. 9C. Observe that this wave is made up of a D.C. component which is proportional to carrier level and an A.C. component which is proportional to the percentage of modulation and the carrier level. If the values of C_D and R_D are properly chosen, this A.C. component will be an exact reproduction of the audio or video intelligence signal. This intelligence signal is fed into a low frequency amplifier (not shown in circuit) for further amplification by coupling the grid of the first low frequency amplifier stage to the D.C. component of the voltage between terminals 3 and 4 is proportional to the level of the modulated R.F. carrier, these terminals may be used as a source for the desired A.V.C. voltage provided that the low frequency component is removed. Application of an A.F. signal to the grid of a controlled tube would place extra modulation on the carrier, a clearly undesirable condition. For this reason, it is necessary to filter the A.V.C. voltage in a radio



Fig. 9. These diagrams, together with the accompanying text, give the complete story of how an A.V.C. voltage can be produced by a simple diode detector, how this voltage can be filtered, and how the voltage is applied to an A.V.C.-controlled R.F. stage.

point \mathcal{S} on resistor $R_{\rm D}$ through blocking condenser $C_{\rm B}$, which blocks out the D.C. voltage component. The circuit is completed through the chassis.

Let us trace D.C. electron flow in the diode detector circuit of Fig. 9A. We start with the diode tube, for we know that electrons flow from the cathode to the plate. These electrons flow through coil L_{\bullet} in the direction indicated by the arrows, then enter terminal S of resistor $R_{\rm D}$, making this terminal negative with respect to the other resistor terminal (4); terminal S is therefore negative with respect to chassis or ground. Furthermore, since receiver in order to keep the A.F. signal voltage out of the A.V.C.-controlled stages.

Filtering the A.V.C. Voltage. The fact that the control grids of R.F. amplifier tubes are negative, so that no D.C. grid current is drawn from the A.V.C. circuit, simplifies the problem of filtering the A.V.C. voltage. In Fig. 9A you will find two A.V.C. filters connected between point 3, at which both D.C. and A.F. components of voltage exist, and point 7 in the grid circuit of an A.V.C.-controlled stage, at which only the D.C. component of voltage is desired. These A.V.C. filters keep the A.F. signal voltage out of the A.V.C.controlled stages of the R.F. amplifier.

Let us consider first the A.V.C. filter made up of C_1 and R_1 . Notice that R_1 - C_1 form a low-pass filter. The series element R_1 offers a far greater opposition to alternating current than C_1 . does, so most of the A.F. is dropped across R_1 . Resistor R_2 and condenser C_2 in the second A.V.C. filter provide additional filtering in the same way, making the voltage at point 7 a practically pure D.C. voltage. Condensers C_1 and C_2 naturally have no effect upon the D.C. voltage, and since no direct current flows through the filter circuit, resistors R_1 and R_2 likewise have no effect upon the value of D.C. voltage at point 7.

The flow of plate current through resistor R_c and condenser C_c in the A.V.C.-controlled stage containing tube VT_1 produces across R_c a D.C. voltage which makes point 6 negative with respect to the cathode; this is ordinary automatic C bias action. The voltage drop across cathode resistor R_c is applied to the grid of tube VT_1 through the chassis path between grounded points 6 and 4, then through R_D , R_1 , R_2 and coil L in turn, thus placing on the grid its normal C bias.

When an R.F. carrier signal is present in the receiver, the D.C. component of voltage produced across $R_{\rm D}$ acts in series with, and aids, the automatic C bias voltage. Thus the A.V.C. voltage and the automatic C bias voltage add together to make the grid of each controlled tube more negative than would be the case without A.V.C. An increase in carrier signal level boosts the D.C. component of voltage across $R_{\rm p}$, driving the grid of each A.V.C. controlled tube more negative and thereby reducing the amplification of each tube sufficiently to keep the signal voltages in all stages of the receiver below the overload values. In an A.V.C. system a condition of equilibrium exists where

the carrier level at the detector is kept just enough above the desired constant value to provide the required A.V.C voltage.

Condenser C_2 in the circuit of Fig. 9A has another important task, that of providing a path to point 6 for the R.F. voltage developed across coil L. If this condenser were omitted, the R.F. current would have to flow through R_2 and C_1 to ground; R_2 would naturally offer considerable opposition to the flow of R.F. current, and there would also be the possibility that R.F. current would stray into circuits where it could cause interference and undesirable feed-back. Since the reactance of C_2 is less than the reactance of the R_2 - C_1 path to ground, R.F. currents will take the C_2 path to ground.

The A.V.C. filter system made up of C_2 and R_2 can be and often is omitted, leaving C_1 and R_1 to do the A.F. filtering and R.F. isolating, particularly when only one tube is being controlled by A.V.C. When several R.F. amplifier tubes are being controlled, it is customary to use an A.V.C. filter similar to R_2 and C_2 in each controlled stage, making connections from each controlled stage to point \mathcal{S} . This serves to isolate the tube circuits from each other, preventing undesirable feedback.

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11

TIME CONSTANT OF THE A.V.C. SYSTEM

An A.V.C. system must prevent blasting when a receiver is tuned suddenly from a weak to a strong signal, and must also compensate for more or less rapid fading effects. For this reason we are interested in knowing exactly how long it takes for the A.V.C. system to get into action when the R.F. carrier level is suddenly changed.

The D.C. component of voltage across R_D in Fig. 9A changes immediately after a change in carrier level,

but it takes a certain amount of time for condenser C_1 in the first A.V.C. filter to charge or discharge to a new voltage value through R_1 .

The amount of time delay introduced by A.V.C. filter R_1 - C_1 depends upon the ohmic value of resistor R_1 and the capacity of condenser C_1 ; this time, when expressed in seconds, is known as the time constant of the A.V.C. filter system, and can be computed quite easily. In the case of Fig. 9A, this can be done by multiplying. the ohmic value of R_1 in megohms by the capacity of C_1 in microfarads. The result will be the time constant of the circuit in seconds, or the time required -for-the-A.V.C. voltage to reach approximately 63% of its final new value after a change in carrier level. (It is standard practice among engineers to specify time constants for 63% of the total change, this having proved more convenient than a time constant based upon a total change.) The A.V.C. filter made up of R_2 and C_2 likewise introduces a time delay, which increases the time constant of the entire A.V.C. filter system. Remember-the time constant of an A.V.C. filter system is determined by the values of the resistors and condensers in the A.V.C. filter system,

A low time constant is naturally desirable in order to make the A.V.C. system respond as rapidly as possible to changes in carrier level; this can be secured by making the values of R_1 , R_2 , C_1 and C_2 low, but doing this impairs the filtering action which is so essential to the operation of an A.V.C. system. Receiver design engineers therefore resort to a compromise which uses filter system parts large enough to provide satisfactory filtering and at the same time small enough to provide a sufficiently short time delay. A time constant of one-fifth to one-tenth of a second for the A.V.C. filter system is considered satisfactory by most engineers for the prevention of blasting and the reduction of fading.

The value for condensers C_1 and C_2 in an A.V.C. filter system have become essentially standard among receiver designers. A capacity of .1 mfd. for C_1 and .05 mfd. for C_2 are generally used, for these condensers are inexpensive and at the same time have a reactance of less than 20 ohms for any I.F. or R.F. signal which may be attempting to flow from resonant circuit L-C into the diode load. A .1 mfd. condenser, when used with a 1 megohm resistor, gives a time constant of onetenth second. The filter action of these parts is such that they will reduce the strength of the lowest audio frequency signal which tries to get into the R.F. and I.F. amplifiers about 100 times. Two of these filter combinations would increase the time constant to one-fifth second and would increase the audio frequency filtering factor to 10,000 times. Ordinarily you will find that the values of resistors R_1 and R_2 range from .1 to 1 megohm, while condensers C_1 and C_2 range from .02 to .1 mfd. Do not be surprised, however, if you occasionally encounter quite different values than these; circuit conditions and the opinions of engineers vary widely. Changing the values of A.V.C. filter resistors or condensers affects the speed of A.V.C. action.

Typical A.V.C. Circuits

Automatic volume controls are usually found only in superheterodyne receivers. As we will soon see, A.V.C. systems reduce the selectivity somewhat, and only superheterodynes have enough to spare. A.V.C. was tried on a few early high-gain T.R.F. receivers, but today, the only T.R.F. sets are inexpensive midgets which do not have this feature.

The sections of a superheterodyne receiver which are usually A.V.C.-controlled are as follows: 1, The R.F. amplifier, which amplifies the incoming modulated R.F. carrier signal; 2, the mixer-first detector, which mixes the incoming R.F. signal with the local ostrolled. If there are three I.F. stages, the second is also A.V.C.-controlled. The final I.F. stage ordinarily has no A.V.C. control, or the control on thisstage may be greatly reduced. This is because each succeeding stage handles greater and greater modulated carrier signal levels, and driving the C bias of a heavily loaded stage highly negative results in amplitude distortion and even partial or complete cut-off of the signal.

The more stages which are A.V.C.controlled, the more uniform will be the receiver output and the less chance there will be for overloading. With these facts in mind, let us examine a



Fig. 10. Simplified diagram showing (in heavy lines) the A.V.C. system of the Philco Model 71 superheterodyne broadcast receiver. The values of the parts are the same as those used by the manufacturer.

cillator signal to give a modulated I.F. signal; *S*, the *I.F. amplifier*.

It is common practice to apply the A.V.C. voltage to all preselector stages, in order to prevent overloading of the mixer-first detector. In those cases where a variable mu or super-control R.F. pentode tube is used for the mixer-first detector, this stage is sometimes A.V.C.-controlled. The first I.F. amplifier stage following the mixerfirst detector is invariably A.V.C.-confew typical A.V.C. systems as used in actual receivers. Since we are interested only in the A.V.C. circuit and the A.V.C.-controlled stages, we will simplify the circuit diagrams by showing only these circuits.

A.V.C.-controlled tubes should always be variable mu screen grid tubes or super-control pentodes, for these tubes give essentially linear amplification over a wide range of C bias voltage values. If ordinary triode, screen grid and pentode tubes were A.V.C.controlled, changes in C bias voltage would cause the R.F. signal to swing beyond the linear region of the tube characteristic, or even beyond the plate current cut-off point, with distortion resulting.

Early Diode Detector-A.V.C. Circuits. One of the first receivers to use a diode tube as a combination demodulator and source of A.V.C. voltage was the Philco Model 71, the circuit diagram of which is shown in simplified form in Fig. 10. Since diode tubes were not available at the time this receiver

The D.C. voltage developed across R_{28} for A.V.C. purposes is filtered for the first I.F. amplifier tube by R_{23} and C_{21} , and is filtered by R_{17} and C_5 for the first R.F. amplifier tube. The time constant of each filter is about .05 second. Notice that both the R.F. and I.F. amplifiers get their minimum C bias voltage from the combination of R_c and C_c in the common cathode lead of the first R.F. tube and the I.F. tube. In this receiver the mixer-first detector has no A.V.C. control. (The *36* tube also acts as an oscillator, so A.V.C. is undesirable on this stage.)



Fig. 11. Simplified diagram showing the A.V.C. system of the Arvin Model 25 superheterodyne euto radio. Notations on the parts are the same as those used by the manufacturer.

was manufactured, the grid and cathode elements of a triode were used, with the plate connected to the cathode as shown in Fig. 10. The diode load, across which exists the A.C. and D.C. components of the rectified detector output, is here made up of two resistors in series, R_{27} and R_{28} , with A.V.C. voltage being taken from the mid-point of the two resistors. R_{28} is a 1-megohm potentiometer with the movable tap connected through a D.C. blocking condenser to the input of the A.F. amplifier, thus providing manual volume control. R.F. currents are kept out of the A.V.C. system and the A.F. amplifier by the filter C_{30} , R_{27} and C_{31} .

Staggered A.V.C. Circuit. An example of a superheterodyne receiver circuit in which less A.V.C. voltage is applied to the amplifier stages near the second detector than to the first R.F. stages is the Arvin Model 25 auto radio circuit, shown in simplified form in Fig. 11. The chief disadvantage of a diode detector, its lack of amplification, is here offset by the use of a type 6B7 duo-diode-pentode tube, with the two diode plates connected together to serve as a single diode detector and the pentode section connected to serve as the first A.F. amplifier stage. This three-in-one tube was created along with the duo-diode-triode to meet the demands of receiver manufacturers who recognized the many advantages of a diode detector in A.V.C.-controlled receivers.

The diode detector load in Fig. 11 is made up of four resistors, R_7 , R_8 , R_9 and R_{10} , connected in series between point 4 and ground. Condenser C_{17} provides a path to cathode for R.F. sigThe mixer-first detector tube gets the lowest A.V.C. voltage, from point 1, while the I.F. amplifier tube gets a greater A.V.C. voltage from point 2. There are no filter resistors in the A.V.C. leads for these last two stages, fast A.V.C. action being desired in preference to complete protection from A.F. signal feed-back.



Fig. 12. Basic circuit for an A.VC.. system in which a separate triode tube is used to provide the A.V.C. voltage.

nals in the detector circuit. The A.C. and D.C. components of the detector output voltage appear across the resistor combination, with points 1, 2, 3 and 4 increasingly more negative with respect to the chassis or ground; when several different values of A.V.C. voltage are provided in this way, the voltages are said to be *staggered*. The A.F. voltage is fed to the control grid of the pentode section from the movable contact of potentiometer R_{11} which, with D.C. blocking condenser C_{16} , is connected across the diode load resistors.

As you can see, the grid of the first R.F. tube is connected to point 3, where it gets the highest A.V.C. voltage of any tube. The filter network for this tube, made up of C_4 and R_1 , has a time constant of .0075 second. This is the only tube in the receiver which has time delay, for fast A.V.C. action is desirable in an auto radio to compensate for changes in signal level when driving. Circuits with Separate A.V.C. Tubes. Before the diode detector came into widespread use, a triode tube in a separate A.V.C. stage was commonly used to provide the required A.V.C. voltage. One basic circuit for this is shown in Fig. 12.

Let us consider the circuit first for the condition where no R.F. signals are being fed to the grid of the A.V.C. tube. Resistors R_1 , R_2 and R_3 form a voltage divider network which is connected across the power pack output terminals, hence electron flow is from B to B+ through these resistors. This makes point x negative with respect to point z, and the grid of the A.V.C. tube (connected to x through grid resistor R_g) is therefore negative with respect to its cathode.

How does the A.V.C. tube secure its plate voltage? Trace from point zthrough the cathode-plate path of the A.V.C. tube to point u, through A.V.C. load resistor R to point w and the chassis, then through the chassis to grounded point y on the voltage divider; this shows that the tube and Rare in series across voltage divider section R_2 , with each getting a portion of the voltage across R_2 . The values of R_1 and R_2 are so chosen by the designer that the A.V.C. tube is biased to cutoff when there is no R.F. signal. Under this condition no current flows through R, making the plate of the A.V.C. tube positive with respect to its cathode by the peaks of the pulses varying according to the modulation signal. The R.F. component of this plate current is filtered out by C_2 and C_3 , while the D.C. and A.F. components appear across R. Electron flow is from u to w through R, hence point u is negative with respect to ground. A.V.C. filters in each controlled stage filter out the A.F. component, while the D.C. component, which varies with R.F. carrier level, is fed from point u to the grids of the con-



Fig. 13. Simplified diagram showing the A.V.C. system of the RCA Model R-74 superheterodyne receiver. This set uses a separate triode A.V.C. tube for producing the A.V.C. voltage. The arrows indicate the directions of electron flow for D.C. Notations on the parts are the same as those used by the manufacturer.

an amount equal to the voltage across R_2 . Furthermore, under this condition points u and w are at the same potential, that of the chassis or ground, and no voltage is fed to the A.V.C.-controlled stages. A cathode resistor R_c is therefore required in each controlled stage to provide normal C bias voltage.

When an R.F. signal acts on the grid of the A.V.C. tube through C_1 , it swings the grid in a positive direction on alternate half-cycles; this allows a pulsating R.F. plate current to flow, with trolled tubes. An increase in carrier level at the grid of the A.V.C. tube increases the D.C. component of voltage across A.V.C. load resistor R just enough to make the grids of the controlled tubes sufficiently more negative to hold the carrier level essentially constant, as in normal A.V.C. action.

Example of Receiver Using Separate A.V.C. Tube. The RCA Model R-74 superheterodyne receiver is an example of a circuit using a separate A.V.C. tube. A simplified diagram of this circuit appears in Fig. 13. First of all, notice that the R.F. amplifier tube, the mixer-first detector tube and the I.F. amplifier tube are A.V.C.-controlled. You can tell this because in each case the grid return lead from the grid coil does not go directly to ground or cathode but to points u and v, which are sources of A.V.C. voltage. The connection is in each case made through an A.V.C. filter consisting of a .05 mfd. condenser and either a 100,000 or 500,-000 ohm resistor.

Resistors R_2 and R_3 in Fig. 13, serving as the load for the A.V.C. tube, divide the A.V.C. voltage into two values, the lower of which is fed to the mixer-first detector and the I.F. amplifier stage. Reduced A.V.C. voltage is necessary here because the mixer-first detector tube must handle a strong local oscillator signal in addition to the regular R.F. signal, and therefore cannot be driven as far negative as the amplifier tubes which get the full A.V.C. voltage.

Now let us see how the A.V.C. tube produces a negative bias voltage which increases with carrier level. We start at the B- terminal of the power pack, knowing that we should be able to trace all D.C. electron flow paths from this point to the B+ power pack terminal, and remembering that electrons flow into the minus terminal of a resistance and out of the plus terminal. One path is from B--- to chassis through choke coil L_{18} , through the chassis to grounded points in the receiver stages, through a cathode bias resistor and through each tube from cathode to plate, and then back to B+ through the plate loads. This complete path through the first R.F. tube is indicated in Fig. 13, with the direction of electron flow shown by arrows and the chassis path indicated by a dotted line; the other paths between ground and the B+ terminal of the power pack,

one for each tube, are all in parallel with this one and can be traced in the same way. Another path is from B through L_{18} to chassis and then through voltage divider resistors R_5 and R_4 directly to B+.

Now we are ready to trace electron flow through the A.V.C. tube (assume for the time being that no R.F. signals are present). Electrons flow from the B--- terminal of the power pack through filter choke L_{17} , then through part of manual volume control potentiometer R_{16} to point z, where the electron flow divides. Some electrons go through the remainder of R_{16} and through resistor R_{17} to ground, from whence they take the various all-inparallel paths through the tubes and voltage divider R_4 - R_5 to the B+power pack terminal, while other electrons go through the A.V.C. tube.

Since electrons enter R_{16} at x and travel to z, x is more negative than zand the cathode of the A.V.C. tube. The grid of this tube, being connected to x through resistors R_{14} and R_{15} , thus gets a definite value of negative bias voltage which makes the tube pass a definite value of D.C. plate current. Electron flow is from z to cathode to plate, through R.F. choke L_9 and A.V.C. tube load resistors R_2 and R_3 to ground, and then to the B+ terminal of the power pack through the chassis and the other tubes in the usual manner. The voltage drops produced across R_2 and R_3 by this current make points u and v negative with respect to ground (point w); these negative voltages, applied to the grids of the controlled tubes through the A.V.C. filters, add to the automatic C bias voltages developed across the cathode resistors of the controlled tubes. There is thus a different normal C bias value for each setting of the manual volume control.

Potentiometer R_{16} serves as a manual volume control, for moving contact arm z closer to x reduces the negative bias on the A.V.C. tube and at the same time increases the plate voltage of the tube slightly. This increases plate current flow, increases the voltage drops across R_2 and R_3 , increases the negative C bias voltage on the controlled tubes, reduces amplification and therefore reduces volume. Moving contact arm z away from x therefore increases volume; with z at t, we have the same arrangement as in the basic circuit of Fig. 12, where the A.V.C. tube has a high enough negative bias to cut off its plate current. Under this condition, it is obvious that no A.V.C. voltage is applied to the controlled tubes, and gain is a maximum.

The plate voltage on the A.V.C. tube is fairly low, for the maximum voltage it can obtain is that which is produced across choke L_{18} by the flow of power pack output current through the D.C. resistance of the choke. As you can see, the A.V.C. tube connects across this choke through R_3 , R_2 L_9 , the z-x section of R_{16} , and L_{17} . The voltage across L_{18} is therefore divided among all these parts and the A.V.C. tube.

When an R.F. signal enters the A.V.C. tube through C_{18} , the tube acts as a C bias detector in rectifying the signal. R.F. components are filtered out by C_{20} and L_9 , while the D.C. and A.F. components of the rectified R.F. signal appear across R_2 and R_3 . The A.F. component is filtered out in the usual manner by the A.V.C. filters. The D.C. component, proportional to carrier level, increases the normal D.C. voltages across R_2 and R_3 and thereby increases the bias voltages applied to the controlled tubes. An increase in R.F. carrier input to the A.V.C. tube produces an increased negative bias for the controlled tubes, reducing amplification just enough to keep the carrier input to the second detector essentially constant.

How A.V.C. Affects Receiver Selectivity. Undoubtedly you noticed in Fig. 13 that the R.F. input for the A.V.C. tube is taken from a point which is two tuned circuits $(L_1-C_1 \text{ and } L_2-C_2)$ alread of the second detector. There is a definite reason for taking the R.F. input voltage for the A.V.C. tube ahead of these highly selective tuned circuits.

In ordinary A.V.C. circuits like those in Figs. 9, 10 and 11, where a diode tube serves both for A.V.C. purposes and for demodulation, we know that when all of the receiver tuning circuits are adjusted exactly to resonance the A.V.C. circuit will vary the gain of the R. F. amplifiers in accordance with the level of the desired carrier. But when the receiver is tuned slightly off the desired carrier frequency, as occurs when a station is tuned in, the carrier level at the A.V.C. input point is greatly reduced by the selectivity of the tuning circuits, and the A.V.C. circuit naturally increases the amplification to offset this. The broadcast thus comes in with about the same volume as before, but is slightly distorted due to incorrect tuning and has a strong noise signal in the background. In the older receivers this was quite objectionable, for listeners were accustomed to tune a receiver according to loudness and this A.V.C. action prevented them from using loudness as a guide. The radio expert would tune a receiver like this for minimum noise, but the average person did not know this little trick and blamed the receiver for the distortion and the apparently poor selectivity.

The solution to this tuning problem involved connecting the A.V.C. tube to a point farther ahead of the demodulator, where there was less sclectivity and where the carrier level for the A.V.C. stage remained fairly constant as the receiver was tuned slightly off the incoming signal frequency. This kept the gain of the receiver fairly constant, allowing the listener to tune for maximum output from the remaining I.F. stages. The R.C.A. circuit in Fig. 13 is an example of this arrangement; with their receiver, loudness could be used as a guide in tuning and good selectivity was obtained.

You will occasionally find the A.V.C. system connected considerably ahead of the second detector in modern receivers; although this practice is desirable it is not entirely necessary. Most receivers are now equipped with tuning aids to indicate when the receiver is correctly tuned.

DELAYED A.V.C.

As you already know, A.V.C. action is undesirable on weak signals, for it reduces amplification a certain amount even on these weak signals. The "trick" which receiver designers use to delay the action of A.V.C. until a definite R.F. input signal (the threshold voltage in Fig. 8) comes through is quite simple, and involves merely the insertion of a fixed D. C. voltage in series with the load resistor of the diode A.V.C. tube.

the plate negative with respect to cathode, and thus no current can flow through $R_{\rm D}$. Any R.F. input voltage $e_{\rm s}$ (secured from the I.F. amplifier) which has a peak value less than the delay voltage $E_{\rm D}$ will not make the diode plate positive, and hence there will be no A.V.C. voltage developed on weak signals. This is indicated in Fig. 14B. where the R.F. signal peak is considerably lower than the delay voltage $E_{\rm D}$ and cannot therefore make the plate positive. In Fig. 14C the peak input signal just equals the delay voltage; plate voltage is thus zero on peaks, but still is never positive. In Fig. 14D, however, the R.F. input signal peak is greater than the delay voltage, and the difference between these two voltages (the shaded area of each pulse) is ef-' fective in sending rectified current through $R_{\rm D}$ for the production of an A.V.C. voltage.

When the positive terminal of the delay voltage source is grounded, as in Fig. 14, this delay voltage places a negative bias on the grids of the controlled tubes at all times. By careful circuit design this can be made to serve



Fig. 14. Basic circuit for delayed A.V.C. action, and curves illustrating how delayed A.V.C. provides maximum amplification for weak signal. E_D represents the D.C. delay voltage, while e_s represents the peak value of the amplified R.F. signal.

The basic circuit arrangement for delayed A.V.C. is shown in Fig. 14A; a separate diode rectifier tube here is used for A.V.C. purposes. D.C. voltage E_D is placed in series with load resistor R_D with polarity as indicated. When there is no R.F. signal input, only this D.C. delay voltage is acting on the plate of the diode A.V.C. tube; it makes as normal C bias for the controlled tubes, eliminating the need for automatic C bias. The ground could just as well be at the negative terminal of $E_{\rm D}$, however. This would put $E_{\rm D}$ in the cathode circuit of the diode, where it would have no direct effect upon the controlled tubes, and would serve only the function of delaying A.V.C. action. A separate diode is a necessity when delayed A.V.C. action is employed to give maximum amplification of weak signals, for with a common detector-A.V.C. tube the delay voltage would prevent demodulation of weak signals and would cause distortion on mediumstrength signals through cutting off of the negative peaks of the A.F. signal.

Delayed A.V.C. Circuit Using Double-Diode Tube. An excellent practical example of delayed A.V.C. is the Silvertone receiver circuit shown in Fig. 15. Section D_1 of the double-diode 6H6G tube serves as second detector, section D_2 serves as A.V.C. tube, and the voltage drop across resistors R_{10} and R_{11} in the power pack circuit provides the delay voltage E_D . Let us analyze the detector circuit action first.

The output R.F. signal e_{s} from the I.F. amplifier is fed directly to diode detector D_{1} through C_{15} , causing rectified current to flow from cathode to plate, through L, R_{4} and R_{5} to the chassis, and then through the chassis to the cathode of D_{1} . The A.F. voltage developed across a part or all of R_{5} is fed through C_{22} to the grid of the 6F5G first A.F. amplifier tube. Incidentally, the grid of this A.F. tube gets its negative bias voltage from the voltage drop across resistor R_{11} (one of the delay voltage resistors).

Now let us analyze the delayed A.V.C. circuit. Condensers C_{14} and C_{15} feed the R.F. signal e_s into diode section D_2 . The plate of this diode is made negative with respect to its cathode (which is grounded) by a connection through load resistor R_7 to one end of resistor combination R_{10} - R_{11} , through which the power pack output current flows to produce a delay voltage. This delay voltage also serves as minimum or normal C bias for the controlled tubes.

Weak signals undergo detection in the conventional manner in D_1 , but cannot overcome the delay voltage which makes the plate of D_2 negative and hence no A.V.C. voltage is developed across R_7 . Strong signals are likewise detected normally by D_1 ; they also make the plate of D_2 positive on the peaks of alternate half-cycles, and electron flow is from cathode to anode of D_2 , then through A.V.C. load resistor R_7 and through resistors R_{10} and



Fig. 15. Simplified diagram showing the delayed A.V.C. system of the Silvertone Model 1986-1987 superheterodyne broadcast receiver (chassis No. 100150). Section D₂ of the double-diode tube produces across resistor R₇ the desired A.V.C. voltage.

 R_{11} to ground and back to the cathode of D_2 . The D.C. component of the voltage developed across R_7 is the A.V.C. voltage; it adds to the normal negative bias voltage E_D and thus amplification is reduced when strong signals come through. The A.F. component of voltage across R_7 is kept out of the controlled tubes by A.V.C. filter R_6-C_8 .

IDENTIFYING A.V.C.-CONTROLLED TUBES .

Ordinarily you will have no trouble in locating the A.V.C. tube on a schematic circuit diagram, for it is now general practice to identify tubes and their functions right on these diagrams. It is not customary, however, to indicate which tubes are A.V.C.-controlled. In order to determine this, you must know the usual methods of applying the A.V.C. voltage to a tube. Three common methods are shown in Fig. 16. You will observe that in each of also as an R.F. by-pass from the tuning coil to ground.

In the I.F. amplifier stages of superheterodyne receivers the tuning condenser is of the trimmer type and is shunted directly across the tuning coil. The circuit arrangement in this case is as shown in Fig. 16B, the A.V.C. voltage still being fed through the tuning coil.



Fig. 16. Common methods of applying A.V.C. voltage to the grids of A.V.C.-controlled tubes.

these circuits the control grid does not trace directly through a conductive path to ground, chassis, B— or C—, but instead traces through an A.V.C. filter resistor to the A.V.C. tube load.

In the circuit of Fig. 16A the A.V.C. voltage is fed to the control grid through the coil (L_1) of the tuned input circuit after being filtered by $R_{\rm F}$ and C_F ; R_c and C_c together furnish the minimum negative C bias for the tube, this being applied to the grid through the chassis and the conductive path of the A.V.C. system (this path is not shown here but can be traced in any of the A.V.C. circuits already studied). With this arrangement, tuning condenser C_1 cannot be connected directly to point x on coil L_1 , because of the fact that the rotor of this condenser is nearly always grounded directly to the chassis. Condenser C_F therefore serves

Occasionally the low R.F. potential end of the tuning coil (point x in Fig. 16C) is grounded, thus giving a direct connection between this coil and the grounded rotor of the tuning condenser. In this case the A.V.C. voltage must be applied through grid resistor R_{g} , with blocking condenser $C_{\rm B}$ used to prevent the A.V.C. voltage from being grounded by tuning coil L_1 . Again the grid traces conductively to the A.V.C. system. To prevent loading of resonant circuit L_1 - C_1 and broadening of its response characteristic, resistor R_s must have a resistance of at least 500,000 ohms. Observe that the cathode is grounded directly; this indicates that the A.V.C. system furnishes the minimum negative C bias voltage in this arrangement. There are two A.V.C. filters, R_F and C_F being the first, and $R_{\rm g}$ with $C_{\rm B}$ making up the second.

Construction Of Manual Volume Controls

Variable resistors or potentiometers which are used as manual volume controls are either of the wire-wound type or of the carbon type. Examples of each type are shown in Fig. 17; as you can see, the general appearance gives no clue toward identifying them.

The total resistance as measured with an ohmmeter between the left (L) and right (R) terminals of a volume control is a rough guide for identifying the type of construction used. Units which have resistances below 5,000 ohms are generally wire-wound: small-diameter resistance wire such as nichrome wire is wound on a thin, long and flexible rectangular strip of fibre and the strip is then curled into a semicircle which fits into the cylindrical metal or bakelite housing of the unit. Insulating material separates the resistance unit from the housing. A movable contact arm, connected to terminal C, provides a means of contact to any point on the winding.



Fig. 17. Two representative manual volume control potentiometers. In each case two nuts are provided on the threaded tubular projection of the housing (through which the sheft runs) to permit clamping the control to a receiver chessis. Only a single mounting hole is needed.

Volume controls which have resistances above 10,000 ohms are ordinarily of the carbon type, although wirewound resistances can be obtained with ohmic values up to 50,000 ohms for special purposes. It is safe to assume that all volume controls above 50,000 ohms are of the carbon type.

It is interesting to know how carbon type volume controls are made. In one type of construction, a thin rectangular strip of insulating material is first coated with a kind of carbon paint, made by mixing highly pulverized carbon particles with water; one mixture of this nature is sold under the tradename Aquadag. The strip is then curled as shown in Fig. 18A and mounted in the metal housing of the control.

The insulating strip is often cut in the shape of a round horseshoe like that shown in Fig. 18B, so the element can be mounted flat against the back of the



Fig. 18. Three types of construction used in cerbon type volume control potentiometers.

housing without curling it. Another scheme involves the molding of a grooved circular horseshoc from bakelite, and pressing into the groove a carbon paste which serves as the resistance element; this type of construction is shown in Fig. 18C.

Each resistor manufacturer generally has his own method of applying the carbon solution or paste and treating it to give greatest dependability during use. Although sliding friction contact arms are used on wire-wound resistors, some form of roller contact which applies direct pressure to the resistance element is generally used with carbon type controls. This is necessary because the carbon elements do not stand up well under constant friction.

Wire-wound volume controls can be made with less than 5% variation in their resistance values (making the resistance of a 1,000 ohm unit anywhere between 950 ohms and 1,050 ohms; engineers call this a 5% tolerance). Carbon resistors cannot ordinarily be made to such close tolerances; a tolerance of 20% appears to be customary with ordinary controls.

Exact values of resistance for volume controls are fortunately not required in ordinary circuits; only where



the volume control is a part of a power voltage-dividing system is it necessary to observe a close tolerance. Variations of from 20% to 40% in the total resistance of a volume control are perfectly satisfactory in ordinary receiver circuits. This is important for you to remember, as you can replace a defective 100,000 ohm volume control with either a 75,000-ohm or 140,000-ohm unit and still secure satisfactory results.

TAPER

The manner in which the resistance of a volume control is distributed is of vital importance, even though the exact resistance of the unit can vary over wide ranges; this distribution of resistance is called the taper of the unit. If the resistance is uniformly distributed, so that varying the movable contact a definite amount causes a uniform change in resistance between points Rand C, we say that the volume control has a linear taper and is therefore a linear control. Thus a linear control set at 1/4 of its total rotation would give 1/4 of the total resistance; similarly, at $\frac{1}{2}$ of the total rotation there would be 1/2 of the total resistance.

 V_{0} olume controls are usually expected to increase the receiver output volume

when the knob is turned in a clockwise direction (making the top of the knob move from left to right). The volume control is usually mounted on the radio chassis in such a way that the shaft, on which the knob is fastened, always points toward the person adjusting the control.

The diagrams in Fig. 19 show the essential elements of a volume control as they are when the shaft points toward you. If the resistance between terminals C and L increases uniformly as the contact arm is moved away from L and toward R, we have a linear control and curve 1 in Fig. 20 will repre-



Fig. 20. Curves showing variation of resistance between terminals L and C with movement for three types of tapered volume controls and for a linear volume control.

sent its variation of resistance with control knob movement or rotation.

Left-Hand Taper. If, in moving contact arm C of a volume control unit from L to R, the resistance between L and C increases slowly at first and then more rapidly and uniformly after the half-way position is passed, we have what is known as a left-hand taper (curve 2 in Fig. 20). The taper or gradual change in resistance is here at the left-hand side of the control.

Right-Hand Taper. If the resistance between L and C in the above case increases more or less uniformly and rapidly at first, and then increases less rapidly as the contact arm approaches terminal R, we have what is known as a *right-hand taper* (represented by curve 3 in Fig. 20). The taper or gradual change in resistance is here at the right-hand side of the control.

There is a simple way of telling whether a particular control has a right-hand or left-hand taper. With the shaft of the control pointed toward you, set the contact arm at its midposition, as indicated in the diagrams in Fig. 19, and then measure the resistance first between terminals C and L and then between terminals C and R with an ohmmeter. If the lower resistance exists between C and L, you have a left-hand taper control, as in Fig. 19A; if the lower resistance exists between C and R, you have a righthand taper control, as in Fig. 19B. If the resistances of the two halves are equal, you either have a linear taper or, in very rare cases, a combination right- and left-hand taper like that represented by curve 4 in Fig. 20.

Why Tapered Volume Controls Are Needed. The main purpose of a tapered manual volume control is to give changes in volume which sound uniform to the human ear for equal changes in volume control position. This means that the receiver designer must take into account the peculiar characteristics of the human ear, which prevent it from detecting changes of less than 3 db in the level of a complex sound. As examples, here are two circuits which use tapered volume controls.

A conventional diode detector circuit in which is incorporated A.V.C. and a manual volume control V.C. is shown in Fig. 21A. A study of this circuit will show you that as the movable contact on the volume control is moved from the cathode end to the diode plate end, the A.F. voltage fed to the triode section of the tube increases. Hence, increased volume is secured when the contact arm is moved towards point R. If a conventional volume control unit is used, it should be so connected that terminal L on it connects to the cathode and terminal R to the tuned input circuit as indicated. This is done to make clockwise movement of the control knob give increased volume.

Supposing that a linear volume control were used for the circuit in Fig. 21A, let us see how it would act. We know that when movable contact C is at R the maximum available A.F. voltage will be applied to the grid of the triode section and full output volume will be obtained; likewise there will be zero A.F. voltage on the grid when C



Fig. 21. Two examples of circuits which require tapered manual volume controls.

is at L, and volume will be zero. It is a known fact that a reduction of onehalf in A.F. voltage corresponds to a 6 db change in sound level. Since the average human ear can just barely detect a 3 db change in the level of a complex sound, the 6 db change would be equivalent to two noticeable changes in sound.

With these facts in mind, let us reduce the volume from maximum to zero by rotating contact arm C gradually from R to L. We listen carefully as we turn; at first we can detect no change in volume—our ear catches one change or reduction—we keep on turning until we can detect another reduction—now we check the volume control knob and find it is at the half-way position, so that half of the maximum A.F. voltage is being used.

We cut the voltage in half again by moving C from the $\frac{1}{2}$ to the $\frac{1}{4}$ position, and again get two noticeable changes in volume or a 6 db change in sound level. Moving C from the $\frac{1}{4}$ to the 1/8 position again cuts the voltage in half, with two more detectable changes in sound. Thus as we move Cfrom R to L, we get two noticeable changes in volume while moving the control knob first $\frac{1}{2}$, then $\frac{1}{4}$, $\frac{1}{8}$, 1/16, 1/32, etc. of its total movement. Clearly a linear control gives extremely non-uniform changes in volume in this circuit, with the changes in volume being most apparent as the contact arm approaches L.

To overcome this non-uniform change in volume with control setting, the radio engineer uses a volume control which reduces the voltage rapidly as C is first moved away from R, then produces less change in voltage with change in setting, or allows the voltage to taper off slowly. This type of volume control would have a left-hand taper. As a general rule, when the current through the entire volume control resistance is constant regardless of the setting of the movable contact, the lefthand taper is required.

Another widely used manual volume control circuit is shown in Fig. 21B. Here resistor R_c sets the minimum C bias for the tube; increasing the resistance of manual volume control resistor V.C. increases this C bias and hence reduces the amplification and receiver volume. To make the volume increase

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with clockwise rotation of the control, the right-hand terminal of the volume control must be connected to resistor R_c as indicated.

Maximum volume is, of course, obtained when contact arm C is at R. As C is moved away from R, the increase in C bias voltage is at first quite rapid, but increases less and less rapidly after that because advancing this volume control reduces the plate current of the tube. It therefore takes larger and larger series resistances in the cathode lead to get appreciable increases in C bias; what we need is a resistor which changes its resistance slowly at first, and then more rapidly. A right-hand tapered resistance, represented by curve 3 in Fig. 20, does this and is therefore used in a circuit like this.

In general, when the current through the volume control resistance changes in value when the movable contact is adjusted, a right-hand tapered volume control is used. In the C bias type of manual volume control such as this, the gain of the tube does not change uniformly with changes in C bias. To compensate for this deviation, special tapers are oftentimes used for the volume control. An ordinary right-hand taper control will generally suffice as a replacement in these cases, however.

You have undoubtedly noticed that the terminals of the manual volume controls in the circuits of Figs. 2, 3, 4. 5, 6 and 7 are marked L, C and R; these markings indicate how the volume control should be connected so as to get increased volume when the control knob is turned in the conventional clockwise direction. In addition, the ohmic values of the controls are given, and each volume control is marked either L.H. to indicate left-hand taper or R.H. to indicate right-hand taper. This extra information will be of practical help to any one who is called upon to service volume control circuits.

Lesson Questions

Be sure to number your Answer Sheet 19FR-2.

Place your Student Number on every Answer Sheet.

Send in your set of answers for this lesson immediately after you finish them, as instructed in the Study Schedule. This will give you the greatest possible benefit from our speedy personal grading service.

- 1. Do receivers which employ automatic volume control also have manual volume controls?
- Give three reasons why automatic volume control is desirable in a radio receiver.
- ✓ 3. Name three types of manual volume controls which are found in radio receivers.
- V 4. What is the most effective way of varying the mutual conductance of a tube in order to control volume?
 - 1/5. What is meant by the threshold voltage in a receiver having delayed A.V.C.?
- 6. What signal voltage (developed by the A.V.C. tube in a radio receiver) is kept out of the A.V.C.-controlled stages by A.V.C. filters?
- $\sqrt{7}$. What determines the time constant of the A.V.C. filter system?
- \bigvee 8. What three sections of a superheterodyne receiver are usually A.V.C.-controlled?
- 9. Which type of taper (left-hand or right-hand) does a volume control unit have if, in moving contact arm C from L to R, the resistance between L and C increases slowly at first and then more rapidly after the half-way position is passed?
 - 10. What is the general rule for using a volume control unit which has a righthand taper?

TAKE THE MIDDLE COURSE

Most of us realize the necessity for moderation in eating and drinking, but we often overlook the fact that moderation in all things is essential to happiness.

Consider, for example, the simple matter of opinions. If a man can see only his own opinions, and is unwilling to recognize that other people may also have good ideas, he is *opinionated*. A man with this fault is often unhappy, because he doesn't get along very well with other people. On the other hand, if a man yields his ideas to another's too readily, he is *weak-kneed*—and also unhappy.

If you can give and take—if you are open to reason—if you steer a *middle course*, you will be liked, people will be comfortable in your company, and you will be following one rule of happiness.

In your dress, be neat but not flashy. In your association with people, be courteous, but not affectedly polite. Be sympathetic, but not sentimental. Be self-confident, but don't be led into difficult situations by overconfidence. Don't believe everything you hear, but don't think everything you hear is false.

Let "moderation in all things" be one of the guiding principles of your life.

J. E. SMITH



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STUDY SCHEDULE No. 20

For each study step, read the assigned pages first at your usual speed, then reread slowly one or more times. Finish with one quick reading to fix the important facts firmly in your mind, then answer the Lesson Questions specified for that step. Study each other step in this same way.

Simple Condenser Filter; Coil-Condenser and Resistor-Condenser Filters; Dual Condenser Filters; Pi or Low-Pass Filters; Practical Examples of Signal Current Filter Circuits; coupling problems; use of D.C. blocking condensers for coupling purposes. Answer Lesson Question 5.

5. Center-Tapped Filament Connections; Analysis of Signal Current Circuits in a Typical Radio Receiver ... Pages 23-29

Interference problems in filament-type tubes; A.C. hum interference; use of center-tapped filament resistor to eliminate hum; schematic circuit diagram of typical 5-tube superheterodyne receiver; tracing signal current paths; locating and identifying extra parts used to keep signal currents in correct paths; analysis of chassis layout to see how effects of stray electric and magnetic fields are minimized by proper placement of parts. Answer Lesson Question 10.

- 6. Mail Your Answers for this Lesson to N.R.I. for Grading.
- 7. Start Studying the Next Lesson.

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World Radio History

How Signal Currents Are Kept in Correct Paths

Introduction

UNDESIRED signal currents can enter a circuit in three different ways: 1, through a <u>conductive</u> path such as a wire or a metal chassis; 2, by <u>electrostatic induction</u>, where the electric field set up by one circuit repels and attracts electrons in another circuit; 3, by <u>electromagnetic induction</u>, where the magnetic field set up by one circuit induces interfering currents in another circuit.

Undesired signal currents are kept out of circuits and desired signal currents are kept in their correct paths by many different methods; although these are generally applied to radio apparatus by design engineers, a knowledge of how each method works is needed by the radio operator or serviceman who is called upon to locate and remedy a defect in some circuit.

The most important of these methods for controlling signal current paths are listed below; many of these have been covered in previous lessons, but this entire subject is of such great practical importance that a more detailed study is entirely justified.

By-Pass Condensers

The importance of by-pass condensers can best be understood by considering the behavior of an amplifier stage which has no by-pass condensers. Such a stage is shown in Fig. 1A, where the input device R_s (across which the input signal voltage e_s appears) and the plate load R_L are shown as resistors. Although tuned circuits and transformers could be used at these locations in practical circuits, under ideal conditions they have the effect of resistors and will therefore be considered as such here.

When the input signal voltage e_s in Fig. 1A is zero, the control grid of the tube is at a fixed potential, the value of

METHODS OF KEEPING SIGNAL CURRENTS IN CORRECT PATHS

- 1. Using by-pass condensers.
- 2. Using R.F., I.F. and A.F. signal current filters made up of resistors, coils and condensers.
- 3. Using D.C. blocking condensers.
- 4. Making proper circuit connections, in order to prevent stray coupling between leads and parts and to eliminate stray currents.
- 5. Shielding signal circuit parts against the effects of stray electric fields.
- 6. Shielding circuit parts against the effects of stray magnetic fields.
- 7. Positioning of power pack and signal circuit parts in such a way that magnetic and electric fields cannot cause interference.
- 8. Twisting filament leads and power supply leads which carry A.C., or running such leads parallel to each other and close together.
- 9. Using center-tapped filament connections for grid and plate return leads.
- 10. Using neutralizing circuits to counteract the effects of R.F. and I.F. feed-back currents. (This last method is covered elsewhere in the Course and will not be repeated bere.)

which determines how much D.C. plate current and D.C. screen grid current will flow. The directions of electron flow in various parts of the amplifier under this condition are indicated by -arrows-directly on the circuit lines in Fig. 1A; you can easily verify these directions by remembering that electrons leave the minus terminal of the B battery (the only D.C. source in the circuit) and travel from cathode to plate or from cathode to screen grid in the tube. The D.C. screen grid and plate currents both flow through resistor $R_{\rm c}$, developing across it a D.C. voltage drop which serves as the normal C bias for the tube and which therefore serves to determine the values of these D.C. electrode currents.

When an input A.C. signal voltage $e_{\rm s}$ is applied across $R_{\rm s}$, it makes the control grid alternately more and less negative than the normal negative C bias value; when this input signal swings the control grid less negative (more positive), D.C. plate and screen grid currents both increase a certain amount, and when the input signal swings the control grid more negative, the D.C. plate and screen grid currents decrease. The application of an input signal on the control grid is thus causing the D.C. plate and screen grid currents to vary continually above and below their normal no-signal values, and we actually have pulsating D.C. currents in these two circuits.

We know definitely that the vacuum tube, when acted upon by the input signal, causes these variations in the electron current which is sent through the screen grid and plate circuits by the B battery; it is quite permissible, therefore, to think of the vacuum tube as simply a variable resistance acting in the plate circuit (and also in the screen grid circuit) and thereby producing these variations or pulsations in plate and screen grid currents. If we say that this tube resistance is varying at exactly the same rate and in the same manner as the grid input A.C. signal, we can neglect the grid circuit entirely and concentrate our attention upon the plate and screen grid circuits, through which the B battery is forcing the pulsating currents.

In an amplifier circuit we are primarily interested in the pulsations or variations in plate current, for these develop across load resistor $R_{\rm L}$ the desired amplified signal voltage. We could consider only this A.C. or signal component of the plate current, and think of it as acting in a circuit made up of the B battery and a varying cathode-to-plate resistance in the tube, but experience has shown that an analysis or study of the circuit under this condition becomes quite complicated.

There is a much easier way of dealing with A.C. or signal currents in the plate and screen grid circuits of an amplifier-a method which is almost universally used by engineers because of its simplicity. This method involves neglecting the D.C. electrode currents entirely, and considering the vacuum tube as our source for A.C. or signal currents. (The battery is neglected as a voltage source and considered simply as a resistor equal to the internal battery resistance $R_{\rm B}$ in Fig. 1A). In other words, we replace the entire grid circuit and the cathode-plate path of the tube with an A.C. generator having a given internal resistance (the A.C. plate resistance) when we deal with the plate circuit, and we likewise consider an A.C. generator in place of the cathode-screen grid path of the tube when dealing with the screen grid cir-(You will recall that this is cuit. exactly what was done in the equivalent tube circuits studied in a previous lesson.)

The tracing of signal currents (A.C. components of electrode currents) in Fig. 1A is quite simple now that we can consider the tube as the A.C. source. A.C. plate current i_p flows through the cathode-plate path of the tube (its source) to the plate, flows through load $R_{\rm L}$, flows to point 2 (the chassis) either through the B battery or through voltage divider R_1 - R_2 , and then returns to the cathode through $R_{\rm C}$; arrows labeled i_p indicate these paths. A.C. screen grid trol grid along with the input signal voltage e_s , and degeneration occurs. What actually happens is this: When the input signal e_s is increasing in a positive direction, making point 3 more positive with respect to point 3 more positive with respect to point 2, the plate and screen grid signal currents through R_c will be increasing. Since electron flow for these currents is from point 2 to point 1, point 2 will be made increasingly more negative with respect to point 1. We thus have acting









FIG. 1B. Three by-pass condensers inserted in the amplifier circuit of Fig. IA, as shown above, effectively serve to keep signal currents in their proper paths, eliminating interfering effects.

FIG. 1C. Pentode R.F. amplifier circuit of Fig. IA with suitable by-pass condensers to provide low-reactance paths for signal currents. Actual tuned input and output circuits are shown instead of resistors.

current flows through the cathodescreen grid path of the tube (its source) to the screen grid electrode, flows to point \mathcal{D} (the chassis) either through R_1 , or through R_2 and the B battery, and then returns to the cathode through R_c ; arrows labeled i_{ag} in Fig. 1A indicate these paths.

Cathode Circuit. Both the plate and screen grid signal currents flow through C bias resistor R_c , producing A.C. voltage drops across it. Since the control grid voltage is developed across R_c , these A.C. voltage drops act on the conon the control grid of the tube (applied to the grid and cathode terminals of the tube) the positively increasing input signal voltage and the negatively increasing A.C. voltage developed across the C bias resistor. These A.C. voltages oppose each other (are out of phase with each other) and hence the net A.C. voltage on the grid is reduced, with *degeneration* or reduction of signal current as a result.

Cathode C Bias Resistor By-Pass Condenser. The method commonly used to keep signal currents out of the

C bias resistor and thus prevent degeneration from occuring in the cathode circuit involves placing a bypass condenser across the C bias resistor, making the reactance of this condenser low enough at the lowest signal frequencies so that the signal currents will take the path through the condenser rather than through the C bias resistor. This is done in Fig.1B, where C_{c} is the cathode by-pass condenser which is connected across C bias resistor $R_{\rm c}$. The A.C. voltage drop across this condenser will be negligibly low because the by-pass condenser has practically no reactance at signal current frequencies, and hence the only A.C. voltage acting on the grid will be the input signal voltage. The D.C. components of screen grid and plate current will continue to flow through $R_{\rm c}$, producing across it the desired C bias voltage.

Screen Grid Circuit. Now let us see what undesirable effects are produced by A.C. screen grid current in the amplifier circuit of Fig. 1A. We note that the screen grid gets its D.C. voltage from a tap at point 4 on the voltage divider network made up of R_1 and R_2 . The voltage of this point with respect to point 2 (B-) depends upon the ohmic values of R_1 and R_2 and upon the current which flows through each resistor to produce across it a voltage drop. From Kirchhoff's voltage law we know that the sum of the voltage drops across R_1 and R_2 must equal the B battery voltage; stated in another way, the screen grid supply voltage (the drop across R_1) will be equal to the B battery voltage minus the voltage drop across R_2 .

Consider the A.C. screen grid current path through R_2 . When a positively increasing control grid voltage causes screen grid current through R_2 to increase, the voltage drop across R_2 will increase and consequently the voltage drop across R_1 (the screen grid voltage) will *decrease*. A decreasing screen grid voltage means a decreasing A.C. plate current, and consequently we have *degeneration*, a reduction in signal output, as an undesirable effect of A.C. screen grid current in this circuit.

Screen Grid By-Pass Condenser. A.C. screen grid current could be kept out of R_2 by placing across it a by-pass condenser (C_x in Fig. 1B) which provided a low-reactance path around this resistor for this current, but a more practical solution is one which places a by-pass condenser between the screen grid lead (point 4) and the cathode lead (point 2); condenser C_{80} in Fig. 1B is connected in this way. Now A.C. screen grid currents have a direct lowreactance path from point 4 to point 2 and then through C_{c} to the cathode, and consequently there will be no degeneration. This practical solution keeps A.C. grid current i_{sg} out of the source as well as out of R_2 .

Plate Circuit. We still have the effects of A.C. plate current to consider in the circuit in Fig. 1A. This current (i_p) flows through the B battery along with the A.C. screen grid current i_{sg} . Any voltage source has a certain amount of internal resistance; this resistance can be considered as acting in series with the source in Fig. 1A, and resistor R_B therefore represents the battery resistance through which these signal currents must flow.

We have seen how screen grid signal current can be made to take a proper path, but we still have plate signal current flowing through this battery resistance and producing a voltage drop across it. By considering the directions of electron flow (indicated by arrows in Fig. 1A for the case where the grid signal is positively increasing) you can see that this voltage drop causes the potential of point 5 (the supply voltage) to decrease with respect to point 2 when the grid input voltage increases. The flow of signal current through the battery thus reduces the plate supply voltage and the plate current, and again we have degeneration. Furthermore, any A.C. plate currents which take the R_1 - R_2 path from point 5 to point 2 instead of going through the battery may also cause undesirable effects.

Radio apparatus today generally contains a number of vacuum tube stages all connected to a common supply source. Any other tubes connected to the battery in Fig. 1A would consequently be affected by the signal voltage drops across the battery resistance and would undergo either regeneration or degeneration, depending upon whether the undesired voltage drops aided or opposed changes in plate current in a particular stage. Clearly it is undesirable to allow signal currents to flow through the plate supply or the voltage divider.

Plate Supply By-Pass Condenser. A by-pass condenser connected directly across the plate supply, like C_P in Fig. 1B, will offer a low-reactance path for signal currents to point 2 (the chassis), thereby keeping them out of the higherresistance paths through the plate supply and voltage divider and eliminating undesirable degeneration or regeneration effects.

The plate by-pass condenser also lowers the impedance of the A.C. plate current path, allowing more A.C. plate current to flow through load resistor $R_{\rm L}$ and thereby developing across it a higher output signal voltage. No useful A.C. plate voltage is wasted now in the internal battery resistance $R_{\rm B}$.

Practical Amplifier Circuit, One example of a practical circuit which requires by-pass condensers is that shown in Fig. 1C, representing a pentode I.F. amplifier stage of a superheterodyne receiver. The connections to the three by-pass condensers are shown by heavy lines to make them easy for you to locate. In this circuit, bypass condensers $C_{\rm P}$ and $C_{\rm SC}$ are both connected to point 2 on the chassis, to which R_c is also connected, and low-reactance by pass condenser $C_{\rm C}$ is relied upon to provide a path from the chassis to the cathode for the signal currents. Oftentimes, however, these by-pass condensers are connected directly from the plate and screen grid circuits to the Design engineers choose cathode. whichever method gives the shortest leads, for long leads result in electric and magnetic fields which produce undesirable coupling between circuits.

BY-PASS CONDENSER RULE

In a vacuum tube circuit, all signal currents are by-passed through condensers to the cathode after they leave the electrodes or circuit parts through which they must flow to give the desired circuit action. Signal currents will take the path through a by-pass condenser in preference to other possible paths between two points in a receiver circuit because the by-pass condenser has practically no reactance at signal current frequencies; in fact, a by-pass condenser is the equivalent of a direct wire connection for these currents.

Occasionally a design engineer omits a by-pass condenser specifically to secure regeneration or degeneration, or uses a single C bias resistor and bypass condenser for two or more tubes.

A few typical examples will fix in your mind the general rule for applying by-pass condensers. A simple triode vacuum tube circuit like that shown in Fig. 2A requires two by-pass condensers. A simple tetrode circuit like that in Fig. 2B requires three bypass condensers. A circuit containing a pentode tube in which the suppressor grid is tied directly to the cathode will likewise need three by-pass condensers. In a pentode tube circuit where the suppressor grid is not at cathode potential, as in Fig. 2C, four by-pass condensers are needed to keep signal curParts in parallel all have the same voltage, and here the resistor and condenser are essentially in parallel with the A.C. generator because the resistance of the connecting wires and the internal resistances of the generators are negligibly small. The condenser has simply increased the load on the A.C. generator and caused more current to be drawn from it, without appreciably lowering the A.C. voltage across the load. Of course, if either generator had a high internal resist-



FIG. 2. Here are three applications of the general by-pass condenser rule. Observe that signal currents in the plate circuit are by-passed to the chassis after they leave the plate load, while screen and suppressor grid signal currents are by-passed to chassis after leaving their respective electrodes. In each case the cathode by-pass condenser provides a low-reactance path from chassis to cathode for signal currents.

rents in their correct paths. Occasionally a vacuum tube is used in a circuit where the control grid is at cathode potential (zero C bias); in cases like this the cathode resistor and the cathode by-pass condenser in the circuits of Fig. 2 would be omitted and the cathode would be grounded.

Simple Signal Current Filters

Simple Condenser Filter. To understand why a simple condenser filter will not always give the desired results, consider Fig. 3A where condenser C_F is connected across load resistor R_L . These two parts are fed from a source which provides both A.C. and D.C. voltage, but we wish to apply only D.C. voltage to the load resistor. The reactance of condenser C_F is low for A.C., but this does not solve the problem, because we still have the full A.C. generated voltage across the load. ance, the increase in alternating current would cause most of the A.C. voltage to be dropped inside the generator, and little would appear across $C_{\rm F}$ and $R_{\rm L}$.

Coil-Condenser and Resistor-Condenser Filters. Let us consider the filter circuit arrangement of Fig. 3B first with only $R_{\rm F}$ connected (CH removed). We see that the A.C. voltage drop across $C_{\rm F}$ is applied across $R_{\rm F}$ and $R_{\rm L}$ in series. If $R_{\rm F}$ is large in ohmic value with respect to $R_{\rm L}$, most of the A.C. voltage will appear across $R_{\rm F}$, and very little across $R_{\rm L}$. However, the D.C. voltage also divides between $R_{\rm F}$ and $R_{\rm L}$ in the same ratio as the A.C. voltage, and $R_{\rm F}$ contributes no filtering action.

To get filtering action and maximum D.C. voltage across load $R_{\rm L}$, we replace resistor $R_{\rm F}$ with choke CH. The choke has a much higher reactance to

A.C. than $R_{\rm L}$, so nearly all of the A.C. voltage is dropped across the choke. For D.C., the choke resistance is low compared to that of $R_{\rm L}$, so most of the D.C. voltage appears across $R_{\rm L}$.

Just as in Fig. 3A, condenser $C_{\rm F}$ in Fig. 3B is ineffective as a filter except when the generators have high internal impedance. We can make this condenser contribute to filtering action, however, by inserting a resistor at point x in Fig. 3B. The alternating current flows through this series resistor and the low combined impedance of $C_{\rm F}$ and $R_{\rm L}$, and most of the A.C. voltsistor $R_{\rm F}$ in place of the choke. Any A.C. voltage across $C_{\rm F1}$ is further reduced by the voltage-dividing action of high-impedance CH (or $R_{\rm F}$) and low-impedance $C_{\rm F2}$, so there is very little A.C. voltage across the load.

Pi or Low-Pass Filters. When the choke coil is used in the circuit of Fig. SC, we have a *pi filter* (so-called because on a circuit diagram it resembles the Greek letter π , which is pronounced "pie") or low-pass filter which can separate low frequency signals from high frequency signals. This filter has a resonant action at a definite fre-



FIG. 3. The actions of the signal current filters commonly used in radio circuits are readily understood by studying these basic filter arrangements. Note the two-circle symbol for an A.C. generator; this symbol is widely used by radio and electrical men.

age drop now appears across the series resistor. The D.C. voltage divides between $R_{\rm L}$ and the series resistor, however, so we may not get enough D.C. voltage across load $R_{\rm L}$ if we use too large a series resistor for filtering purposes.

If high D.C. load voltage and adequate filtering are both required, a choke must be used at x in Fig. *SB* instead of a series resistor. This gives two choke coils and one condenser in the filter network, and is a highly effective arrangement because the chokes offer high impedance to A.C. but very little opposition to D.C.

Dual-Condenser Filters. In practical circuits the source resistance is usually appreciable, and here the filter arrangement shown in Fig. SC is highly effective. It has only one choke coil, and will work fairly well even with requency known as the *cut-off frequency;* signals below this cut-off frequency pass through the filter readily and enter the load, for the choke coil then has very little reactance and the condensers have such high reactances that there is little shunting effect on signals. At frequencies slightly above the cut-off value, resonant action prevents transfer of signals through the filter. Signals considerably above the cut-off frequency are blocked by the choke coil and passed by the condensers, and hence cannot reach the load.

The cut-off frequency value depends upon the inductance of the choke and the capacities of the condensers (which are usually of equal size). The cut-off frequency is easily found by considering the filter as a resonant circuit made up of an inductance equal to one-half the inductance of the choke coil and a capacity equal to that of one of the condensers in the filter, then determining the resonant frequency by the usual procedure for resonant circuits.

Practical Examples of Signal Current Filter Circuits. The tuned radio frequency amplifier circuit in Fig. 4A contains two examples of the resistorcondenser filter shown in Fig. 3B. Condenser C_{F1} provides a direct path to cathode for grid circuit R.F. currents, while the C_{F1} - R_{F1} filter combination prevents any A.C. voltages which may exist across R_c from affecting the grid of the tube (there may be small power supply and prevents any alternating current in the power supply from getting into the plate circuit. Filter resistor R_{F2} naturally lowers the D.C. plate voltage a certain amount; if this reduction in voltage is undesirable, this resistor may be replaced by an R.F. choke coil. Unfortunately, this substitution of a choke coil only serves to prevent plate circuit signal currents from getting into the power supply; the choke coil is not effective in keeping low frequency power pack ripple currents out of the plate circuit unless an additional iron-core choke





Fig. 4. The signal current filter is shown by heavy lines in each of these radio circuits.

amounts of A.C. plate current and power pack ripple current flowing through $R_{\rm c}$ and developing A.C. components of voltage across it). What happens is this: C_{F1} , having a very low reactance, is in series with R_{F1} , a very high resistance, across $R_{\rm c}$, so what little alternating current gets through $R_{\rm F1}$ produces only a negligible A.C. voltage drop between points 1 and 2 as it flows through $C_{\rm F1}$. Only the desired steady D.C. voltage across $R_{\rm c}$ can act on the grid through R_{F1} . Remember this important fact: A signal current filter will act both ways, preventing signal currents which are in one circuit from getting out and preventing other alternating currents from getting in.

In the same manner, signal current filter C_{F2} - R_{F2} prevents plate circuit signal current from getting into the designed for this particular purpose is used in series with the R.F. coil. For effective two-way filter action, R_{F2} should be replaced by R.F. and A.F. choke coils in series. Because of the relatively high cost of choke coils, it is more economical to use a filter resistor and either increase the D.C. supply voltage or adjust the circuit to operate satisfactorily at a reduced value of plate voltage.

The diode detector circuit shown in Fig. 4B contains a practical example of the dual-condenser filter represented by Fig. 3C. The path taken by the rectified signal current is through the diode tube, input coil L, filter resistor $R_{\rm F}$, and diode load resistor $R_{\rm L}$. R.F. current is kept out of the diode load by the filter combination made up of $C_{\rm F1}$, $R_{\rm F}$ and $C_{\rm F2}$. In some circuits you will find that C_{F2} is omitted, and the wirc going from point 1 to the A.F. stage through blocking condenser C_B is surrounded by a grounded metal shield. This shield acts as a capacity to the chassis, and thus serves the double duty of substituting for condenser C_{F2} and shielding this lead from stray electrostatic and magnetic fields.

I have already pointed out that choke coils may be used in place of filter resistors. One important objection to the use of choke coils is the fact that their characteristics change greatly with the frequency. Wherever cathode through by-pass condensers $C_{\rm FS}$ and $C_{\rm C}$. The cut-off frequency for this filter can be any value lower than the I.F. value but higher than the highest modulation signal frequency being handled. The signal current filter made up of $C_{\rm F3}$ and $R_{\rm F}$ in Fig. 5A serves to keep signal currents in the plate signal circuit and to kccp power pack ripple current out of the plate signal circuit.

An I.F. amplifier circuit using a pentode tube and two resistor-condenser signal current filters is shown in Fig. 5B. You can readily identify conden-



FIG. 5. Triode second detector circuit (A) and I.F. amplifier circuit (B) containing examples of signal current filters (shown in heavy lines).

possible, non-inductive carbon or metallized resistors are therefore used as filter resistors, particularly in highfidelity audio amplifiers and in the preselector stages of all-wave superheterodyne receivers, for their effects are the same at any frequency.

The second detector circuit in Fig. 5A is an example of a superheterodyne receiver stage where a pi or low-pass filter is used. The combination of C_{F1} , CH and C_{F2} forms a filter like that in Fig. SC, which definitely suppresses the I.F. signals while allowing direct current and low or audio frequency current to pass through. The desired modulation signal currents thus pass through load $R_{\rm L}$ and return to the

ser C_{c} as the by-pass condenser for automatic C bias resistor $R_{\rm C}$; this condenser also provides a low-reactance path from ground to cathode for all signal currents. You will also recognize combination R_1 - C_1 as the A.V.C. filter for this stage; this filter has a two-way action, providing a path to ground for R.F. grid currents (thus keeping them out of the A.V.C. system) and preventing the A.C. components of A.V.C. voltage from entering the grid circuit. Filter R_3 - C_3 forces the A.C. plate current to go to the chassis and then through C_{c} to the cathode instead of entering the power pack. Resistors R_2 and R_4 serve primarily as a voltage divider which provides the screen grid with the correct portion of the supply voltage, but R_2 also serves with C_2 as a filter which prevents screen grid signal currents from flowing through the supply lead and the power pack to the cathode. C_2 provides a direct path from screen grid to cathode for these currents.

D.C. Blocking Condensers

When the plate circuit of one stage is to be coupled to the grid circuit of a following stage and both stages have common supply terminals, the coupling device must be of such a nature that it keeps D.C. supply current out of the grid circuit of the following stage. (This discussion does not apply to direct-coupled amplifiers.) If this precaution were not observed, the grid would become positively charged with respect to its cathode, and correct operation would not be secured. When radio or audio frequency transformers are used as plate loads, the signal is transferred from one stage to another by induction and the D.C. component of plate current can flow only in the primary of the transformer, effectively solving this problem. When resistors or choke coils are used as plate loads, however, special means must be used to prevent D.C. supply current from entering the following grid circuit.

A universally used solution to this coupling problem is that shown in Fig. 6A, where coupling condenser C_B provides a low-reactance path for signal currents between the two stages and effectively blocks any flow of direct current. The currents flowing through each part in this circuit are indicated. You can readily see that if there were a D.C. conductive path for electrons from point 2 to point 1, electrons starting from point 3 (which is at ground or B— potential) would pass through grid resistor R_{g2} to point 2, and then to point 1 and through resistors R_{P} and R_{F} to the B+ terminal of the power pack, making point 2 positive with respect to ground (point 3) and thereby placing a high positive bias on the grid of the following tube.

Quite often an external D.C. voltage is to be applied to a tube electrode which is already connected to ground by a D.C. conductive path. The R.F. amplifier circuit in Fig. 6B is an example; oftentimes an A.V.C. voltage must be applied to the grid of the tube, yet because the rotor of the tuning condenser is permanently grounded to the chassis and at the same time connected to one side of low-resistance coil L_1 , an A.V.C. voltage applied directly to the grid would be shorted to ground through L_1 .

One solution to this problem appears in Fig. 6C. The A.V.C. voltage is applied to the grid of the tube through an A.V.C. filter in the usual manner, with resistor $R_{\rm g}$ inserted as shown to prevent the A.V.C. circuit from shunting to ground (through A.V.C. filter condenser $C_{\mathbf{F}}$) the R.F. signal voltages produced across tuning circuit L_1 - C_1 . The value of $R_{\rm g}$ may be several hundred-thousand ohms; since the grid of the tube is negative at all times, it draws no current, and any currentlimiting effects of resistors in the grid circuit are unimportant. In addition, D.C. blocking condenser $C_{\rm B}$ is required to prevent shorting out of the A.V.C. voltages by coil L_1 . This is known as the shunt feed method of applying an A.V.C. voltage, for the A.V.C. voltage acts in parallel or in shunt with the signal input voltage.

Another solution to this same problem. used when the coil can be disconnected from the chassis, is the *series feed* method of applying the A.V.C. voltage, shown in Fig. θD . The A.V.C. lead is connected to the lower end of coil, L_1 , through resistor R_F of the A.V.C. filter. A.V.C. filter condenser C_F serves as a D.C. blocking condenser in preventing the A.V.C. voltage from being shorted to ground, and at the same time provides the required lowstray currents in ganged variable condensers, for each of these defects can cause either regeneration or degeneration under certain conditions.

An engineer can sit down at a drawing board and, after making certain engineering calculations, design a radio receiver or an amplifier which has cer-



FIG. 6. Examples of circuits which require and use D.C. blocking condensers.

reactance path from the coil to ground for R.F. signals.

Importance of Proper Connections

The use of filter circuits to keep R.F., I.F. and A.F. currents out of the power supply is in itself no assurance that regeneration or degeneration effects will disappear. We still have to consider the problems of stray coupling between leads, common coupling between different circuits due to stray currents in the metal chassis, and common coupling between circuits due to tain desired characteristics. It is quite another matter, however, to convert this design into an actual model which has these same characteristics. Regeneration and degeneration effects generally give the most trouble, making necessary considerable experimentation in order to determine the best locations for the various parts and connecting wires.

Long connecting wires from the grid to the grid input circuit, from the plate of a tube to the plate load, or from the screen grid electrode of a tube to its power supply terminal are to be avoided, for they can give trouble in the form of capacitive or inductive coupling between each other.

Quite often a receiver manufacturer will allow a certain amount of regeneration or degeneration to exist because it does not appear to be objectionable at the time, but will later be compelled to modify certain sections of the receiver because the undesired effects become too objectionable after the receiver has been in use for some time. This is why you will occasionally find receivers of the same model, but made at different times, with minor changes connections and parts values. in Furthermore, since Radiotricians are called upon to make these corrections on earlier receivers which have developed trouble, the practical importance of this problem of making proper connections is quite evident.

Proper Connections in a Single R.F.Stage. As a practical example, let us consider how connections should be made in a typical pentode R.F. amplifier stage, such as that represented by the schematic circuit diagram in This diagram tells 7A. Fig. us that A.V.C. filter R_1 - C_8 is used to keep A.F. signals out of the grid circuit, to delay the A.V.C. action, and to provide a low-reactance path from the grid circuit to the chassis for R.F. signals. By-pass condenser C_4 provides a low-reactance path for signal currents around C bias resistor $R_{\rm c}$, while screen grid filter R_2 - C_5 keeps screen grid signal current out of the power supply; plate supply filter R_3 - C_6 does the same thing for plate circuit signal currents. This diagram does not show, however, how these signal currents actually travel between points 1, 2 and 3 through the chassis to point 4. The paths taken by these currents are quite important, for degeneration or regeneration may occur if the signals from the screen grid, plate and control grid get a chance to produce appreciable voltages and mix before they reach the cathode. (Although the signals mix in passing through C_4 , the reactance of this condenser is so low that the undesirable voltages produce negligible regeneration or degeneration effects.)

The production design engineer, who must decide beforehand the best location for each connecting lead and for each part, might redraw the schematic circuit diagram of Fig. 7A in the manner shown in Fig. 7B in order to indicate that points 1, 2 and 3 be connected directly to point 4. Doing this prevents the screen grid, control grid and plate signal currents from wandering through the chassis and mixing together before reaching point 4. All by-pass condensers in a single R.F. stage should be connected to a common point in the stage in order to prevent undesirable direct coupling between the different circuits in the stage.

The arrangement of connections shown in Fig. 7B may insure freedom from troubles due to improper direct connections, but we can still have trouble due to capacitive or inductive coupling between leads. Grid and plate leads in an R.F. stage must be kept as short and as far apart as possible in order to prevent capacitive and inductive feedback of signals from the plate circuit to the grid circuit. Suppose that, through faulty chassis layout, plate lead z is placed close to control grid lead x; this would allow signal feedback from the plate circuit to the grid circuit, causing either regeneration or degeneration. Keeping the input and output leads in their proper places, as far away from each other as possible.

is a highly important duty of the radio apparatus manufacturer; since the exact positions of connecting wires are seldom shown in radio receiver service manuals, it is essential that you be familiar with approved methods of making actual connections.

Actual chassis connections for the pentode R.F. amplifier stage of Fig. 7A

point; in fact, one is above the chassis and the other is below.

Effects of Stray Chassis Currents. Suppose that instead of making connections as shown in Fig. 7C, points 1, 3 and 4 were simply connected to the nearest convenient points on the chassis, as indicated in Fig. 7D. Now we would expect the grid circuit signal



FIG. 7. An ordinary pentode I.F. amplifier circuit appears in its simplest form in the schematic diagram at A. Diagram B shows actual connections of by-pass condensers to a common point, but still in schematic form. Diagram C shows actual connections of parts on the chassis, while D shows how improper grounding of by-pass condensers to the chassis can result in interfering currents.

are shown in Fig. 7C, as they would appear when looking at the bottom of the chassis. The input and output coils with their trimmer condensers are, of course, on the top of the chassis, as also is the connection from the input tuned circuit to the top cap of the tube. Take particular notice of the fact that the control grid lead x and plate lead z do not run close to each other at any current i_s to flow from point 1 through the chassis to point 4, and A.C. plate current i_p to flow from point 3 through the chassis to point 4. This is what actually takes place, but the A.C. plate current, which is many times greater than the grid current, spreads out over a wide path through the chassis in traveling from point 3 to point 4. One possible curved path is indicated in
Fig. 7D; notice that now some of the A.C. plate current i_p is flowing over the same path as the grid circuit signal current i_g . The A.C. plate current which takes the curved path causes an additional A.C. voltage drop between points 1 and 4 in the grid circuit, and this may cause undesirable regeneration or degeneration, depending upon the phase relationship between the signal currents involved.

If the grid by-pass condenser in a following stage is connected by mistake to a convenient point on the chassis, this signal current can likewise wander through the chassis, with one of its many possible paths being from point 1 to point 4 in Fig. 7D. This current can likewise cause undesirable regeneration or degeneration, depending upon the phase relationships of the currents.



Although you ordinarily think of power pack filter circuits whenever electrolytic condensors are mentioned, here is an electrolytic condenser which is widely used to keep audio signal currents in correct paths. It is a lowvoltage, high-capacity tubular dry electrolytic, and is used to by-pass a.f. currents around cathode resistors in the a.f. amplifier stages of radio receivers.

If it is desirable from a production viewpoint to connect points 1 and 3 directly to the chassis, as indicated in Fig. 7D, a length of heavy copper wire running from point 1 to point 4 and then to point 3 will provide a single path for all these A.C. currents, keeping them from straying through the chassis. Since this heavy wire will have low resistance, the current will take this path in preference to the highresistance path through the metal chassis. You will occasionally find this practice followed in radio receivers; a single heavy bus wire (copper wire having a square cross-section) is run from point 1 to point 4 to point 3 of one tube to point 1 to point 4 to point 3 of the following tube and in turn to each following tube. The connection must be made in this logical order if interference between currents is to be prevented.

Stray Currents in Ganged Variable Condensers. The ganged variable condenser in a sensitive modern all-wave receiver will generally have three tuning sections, with the rotors grounded (theoretically, at least) to the chassis through the frame of the condenser, as shown schematically in Fig. 8A. There will be a connection from each stator section to the grid of a tube and to a tuning coil.

With the stator sections so close to each other on the condenser unit, you can readily see how signals in one section might affect an adjacent stator section, causing feed-back effects. To eliminate this possibility, shielding plates are usually placed between the stator sections, these plates being grounded to the frame of the condenser and therefore to the chassis. (Electrostatic shielding of this type will be taken up later in this lesson.)

The average technician fails to realize that the current in each resonant circuit must flow through the rotor shaft in order to get to the condenser frame and then to the chassis. The construction of the rotor plate assembly is such that this path has appreciable resistance, and a certain amount of regeneration or degeneration occurs when different signal currents flow through the same section of the rotor. In the circuit of Fig. 8B, for example, signal currents through condenser section C_2 can either flow through the apparent rotor resistance R_2 of this condenser and then through rotor resistance R_3 of condenser C_3 to ground, or can simply flow through the rotor resistance R_1 of condenser C_1 to ground; in either case there will be interference between signal currents. In order to eliminate this trouble, it is necessary to provide low-resistance paths to ground for the signal currents of each making the stray currents appreciable in relation to desired signal currents. Rather elaborate precautions must often be taken to eliminate these stray currents; a study of the gang tuning condensers in a few modern all-wave receivers will reveal the exact manner in which low-resistance paths to ground are provided for each rotor section.

Shields for Electric Fields

Having seen how interfering currents can get into circuits by actually



FIG. 8. Connections for a typical three-section ganged variable condenser appear at A. The effective locations of rotor shaft resistances are shown at B (curved arrows represent rotors), while one method used to offset the effects of rotor resistance and eliminate stray currents is shown at C. R_1 , R_2 and R_3 are not actual resistors, but represent resistances existing in each rotor section because of its mechanical construction.

rotor section. Thus you will often find wiping contacts on each side of a rotor section, with each pair of contacts connected together and also connected directly to the chassis, as shown in Fig. SC. A better connection is from the rotor wiping contacts directly to the low or grounded R.F. terminal of the tuning coil associated with a rotor section.

Since the currents flowing in the rotor sections are resonant stepped-up currents, they may often be quite large, flowing over undesired conductive paths, we are ready to consider how interfering currents can invade a circuit even when there are no direct wire connections. When the medium by which the interfering signals enter is an electric field, we call the action <u>electrostatic induction</u>, and when a magnetic field is to blame, we call it <u>electromagnetic induction</u>. We will consider electric fields first.

How Electric Fields Affect Grid Circuits. The grid or grid lead of a vacuum

tube is particularly susceptible to stray electric fields, for extremely small induced A.C. grid voltages can produce strong A.C. plate currents. The action is as follows: When the grid lead of a vacuum tube is located in the electric field which surrounds a highly charged object, variations in the electric charge (potential) of the object will cause the electric field to vary correspondingly. and this varying electric field will influence the free electrons in the grid circuit. When the object is negatively charged, it will repel the grid electrons; when the object is positively charged. it will attract the grid electrons (like

tion or in connecting leads. There will also be an interaction of electric fields when plate leads run near grid leads. These are just a few typical examples of how interfering currents are produced by stray electric fields. These interfering currents can result in regeneration, in degeneration or even in undesirable modulation of an R.F. signal, depending upon circuit conditions. Excessive regeneration, which results in a hissing noise in the loudspeaker or even in squeals due to oscillation, is easily recognized as being due to undesired coupling between circuits. Likewise, a 60 or 120 cycle hum modula-



FIG. 9. These disgrams show how an electric field can affect electrons in the grid circuit of a tube, and how a metal plate serves as a shield for electric fields or electric lines of force.

charges repel, and unlike charges attract). If the object is being charged by an A.C. potential, the free electrons in the grid circuit will move back and forth to produce an alternating current of the same frequency.

Sources of Electric Fields. The plate of a vacuum tube is one object which can have a varying potential and produce an electric field which affects other circuits. The electric field associated with a power transformer or audio transformer can also induce currents in nearby tubes and in conductors. The stator plates of one section of a variable condenser can produce varying electric fields which induce interfering currents in another stator section can be recognized and often traced to undesirable coupling between a power supply part and some signal circuit.

Shielding of Coils. The electric field around a coil which is carrying a current can be represented as shown in Fig. 9A. When this electric field passes near the control grid lead of a vacuum tube, the electrons in this lead will move back and forth in accordance with variations in the current through the coil, and consequently an interfering current will be induced in the control grid lead.

If a metal plate is inserted between the coil and the exposed grid lead of the tube, as indicated in Fig. 9B, this metal plate will serve as a shield which prevents the electric fields (lines of force) from going any farther. Only materials which are good conductors of electricity will serve as shields for electric fields; insulating materials have little or no blocking effect upon an electric field.

The effectiveness of a shield in suppressing an electric field is easily demonstrated if you have access to a radio receiver which does not have A.V.C. Couple the receiver to the antenna system in the manner shown in Fig. 9C, so that the only connection to the antenna is by capacity coupling beelectric shield in radio apparatus.

Practical Data on Electric Shields. The use of metal plates as shields between interfering parts is an old idea in radio, and this practice is followed even today in isolating the sections of many ganged variable condensers. Metal plates are entirely impractical, however, for shielding other parts in a receiver, as an electric field radiates in all directions and can affect many parts simultaneously.

The solution to the problem of suppressing electric fields lies in surrounding with a metal case the coil or other part which produces the field in order



FIG. 10. Examples of electric shields used for radio apparatus.

tween metal plates a and b. With plate a removed, tune the receiver to a station which can just barely be heard. Now as you bring plate a near plate b, the volume will increase. Inserting a third metal plate S, larger in size, between plates a and b without touching them will have no effect on volume, for plate S merely relays the electric charges from plate a to plate b. When plate S is grounded, however, the electric field set up by plate a is grounded through S, and the volume will drop to the point where it is just barely audible. This little demonstration shows clearly the importance of grounding an to prevent spreading of the field. Parts which may be affected by electric fields are surrounded with metal shields in much the same manner. Thus, grid leads are surrounded by flexible metal sleeves (with insulation between the sleeve and the grid wire to prevent a short circuit), metal shields are used over glass tubes, or tubes which have all-metal envelopes are used to overcome the effects of electric fields.

Typical metallic shields which have been found effective in preventing interference by electric fields are shown in Fig. 10. The metal plates used between sections of a ganged variable condenser are illustrated at A. Two types of metal shields for glass tubes appear at B. At C is shown a shielded wire; an ordinary insulated wire is covered by a flexible metallic braid or metallic loom, and the metal is in turn covered by a layer of insulation. Sometimes this outer layer of insulation is omitted, and sometimes there may be two or more insulated wires inside the metallic loom.

In addition to supporting the various parts, the chassis of a radio receiver also serves as a shield which is effective in preventing interference between parts above the chassis and those below the chassis; a typical chassis made of heavy gauge sheet metal is shown at D.

An example of an aluminum shield used for R.F. coils and R.F. transformers appears at E. In all cases it is essential that the shield be grounded to the metal chassis of the receiver at several points.

Shielding is so important to the operation of a high-gain, high-quality radio receiver that totally exposed parts (except resistors) are becoming quite rare. Paper condensers are automatically shielded if that lead marked "outer foil," "grounded end," or "ground" is connected to the chassis; a great many types of electrolytic condensers are made with metal shields, these being automatically grounded by bolting the unit to the chassis. The envelopes of all-metal tubes are likewise grounded automatically by inserting the tube in its socket, for one prong of the tube is always connected to the metal envelope, and the socket terminal for this prong is usually grounded.

Electromagnetic Shields

Stray magnetic fields, which are produced by coils or wires carrying varying electric currents, can induce interfering voltages in other coils or wires. These induced voltages are especially troublesome in grid circuits.

Two distinct electromagnetic shielding problems arise in the construction of radio equipment: 1, the elimination of interference due to parts which carry low-frequency power supply currents or A.F. signal currents; 2, the elimination of interference produced by parts which carry R.F. or I.F. currents.

Shields for Low-Frequency Magnetic Fields. In the case of magnetic fields which vary at a low frequency, the solution is simple. The part in question, usually an A.F. or power line frequency choke or transformer, is placed in an iron or steel housing which is so designed that leakage magnetic fields will flow through this housing and return to their correct paths instead of straying through the receiver and making trouble. Surrounding a coil with iron or steel lessens the temptation for lines of force to take highopposition paths through air; iron or steel housings or shields are therefore widely used for low frequency coils and transformers.

Shields for High-Frequency Magnetic Fields. At frequencies above about 50 kc., it is not necessary to use an iron or steel housing for shielding purposes; in fact, materials with good conductivity, such as aluminum and copper will give better shielding effects than poor-conductivity iron and steel.

You can perform a simple experiment which will show how R.F. magnetic fields are affected by shields. Two coils, L_1 and L_2 , are connected as shown in Fig. 11. Any air-core coils will do; the numbers of turns are not important; Coil L_2 is connected to the antenna and ground terminals of a radio receiver which does not have A.V.C. and which is tuned to a strong local station. Coil L_1 is connected to the antenna and ground, as indicated. Antenna current flowing through L_1 sets up a varying R.F. magnetic field which induces an R.F. voltage in coil L_2 , and signals are therefore transferred to the receiver input.

Now insert between coils L_1 and L_2 a reasonably large copper, brass or aluminum plate, or place an ordinary aluminum cooking pot or frying pan between the two coils. You will find that this prevents transfer of the magcuits produce magnetic fields to oppose the original magnetic fields; this is simply Lenz' Law, with which you are already familiar. The opposing magnetic flux repels the original flux, sending it back and thus preventing the original flux from penetrating the metal shield. The circulating currents produced in the metal shields are called eddy currents. Of course, a certain amount of power is wasted when eddy currents flow in a conductor, but if the shield is not placed too close to the strongest part of the magnetic field, the loss will be negligible.



FIG. 11. In this experiment, a conductive metal plate inserted between the two coils prevents R.F. signals from reaching the receiver.

netic field from one coil to the other, and therefore prevents the local station from being heard. If this metal plate is grounded, it will also act as a shield for electric fields.

The action of the metal shield in Fig. 11 can be explained as follows: In attempting to pass through the conductive metal plate, the varying magnetic flux sets up very small circulating currents in the metal plate. You can consider this plate as being made up of many small circular metal rings, each of which is a complete closed circuit for electrons. The varying magnetic flux induces a voltage in each of these closed circuits in such a manner that the currents flowing through the cir-



FIG. 12. How eddy currents in a metal plate can set up an opposing flux.

High-grade R.F. and I.F. coils are usually placed in aluminum or copper cans, but copper-plated steel housings are also used as shields for coils. Aluminum and copper are good conductors of electricity, and therefore allow larger eddy currents to flow; these currents in turn produce stronger opposing magnetic fields to keep the original flux from passing out of the can. By placing the shield a reasonable distance away from the coil (at least one-half the diameter of the coil), little loss of signal power is incurred. Since the shield in effect acts as a larger number of short-circuited secondary windings on the coil, the main flux of the coil is reduced and the inductance of a shielded coil is therefore lower than that of an unshielded coil. The relationship of the original flux, the eddy current rings and the opposing flux are indicated in Fig. 12. Typical shielded coils are illustrated in Fig. 13.

Positioning of Parts for Minimum Interference

When shielding of parts is impractical or is not completely effective, inter-



FIG. 13. Examples of shielded R. F. coils.

fering effects can be kept at a minimum by proper positioning of those parts and wires which are most affected. A

the vicinity of a coil is indicated in Fig. 14. For purposes of simplicity, an iron-core choke coil having open core ends was chosen. When a wire which is a part of a closed electrical circuit is placed in this magnetic field in such a way that the field cuts across the wire (or loops through the closed circuit formed by the wire), as is the case for wire a in Fig. 14, a voltage will be induced in the wire. When the wire is placed in position b (Fig. 14), the magnetic field will be parallel to the wire instead of cutting it or linking through the closed circuit, and no voltage will be induced. Thus you can see that for minimum interference, a wire should be parallel to the magnetic lines of force in its vicinity.

When a coil is placed in the magnetic field produced by the iron-core choke in Fig. 14, the relative positions of the two coils will determine whether or not a voltage is induced. For example, coil A in Fig. 14 is placed with its axis parallel to that of the choke coil, and hence the magnetic field links through coil A,



FIG. 14. The position of a coil or wire in a magnetic field determines whether or not a voltage will be induced.

review of the nature of magnetic fields and their voltage-inducing characteristics will reveal the best positions for various parts and wires.

The nature of the magnetic field in

producing an induced voltage. Observe that the magnetic lines of force which pass through coil A are parallel to the axis of the coil. The axis of coil B, however, is at right angles to the magnetic lines of force; in this case no flux flows through the turns of the coil and consequently there is no voltage induced.

Positioning of Wires. When it is necessary to place two wires close together, and one of the wires carries a varying current which can produce a varying magnetic field and in turn induce an interfering current in the other wire, the two wires should be placed at right angles to each other in the manner shown in Fig. 15A. Doing this makes the magnetic lines of force produced by wire x cut across wire y twice in opposite directions, with the result that the

paths are at right angles to each other. as indicated in Figs. 15B and 15C, and should be as far apart as possible from circuits which are most affected. This rule is, of course, only approximate, for the fields produced by these devices may be considerably distorted. Final positions for minimum interaction can be determined by experimentation; in the case of audio and power transformers, this can be done by feeding an A.F. or power line voltage to one coil and connecting a pair of headphones either directly across the other coil or through a separate audio amplifier which boosts the voltage and gives



Fig. 15. Interfering currents due to electromagnetic induction are a minimum when wires, iron-core devices and coils are positioned at right angles to each other, as shown here.

two voltages induced in y cancel each other and have no effect upon the circuit of which wire y is a part. For example, one line of force produced by wire x may cut wire y at point 1 at a certain instant, producing in y an induced voltage having a certain polarity, but at the same instant this line of force will be cutting wire y in the opposite direction at point 2, producing a voltage of opposite polarity at 2. Placing wires at right angles to each other in this way also reduces interaction of electric fields.

Positioning of Coils. Coils and transformers should be positioned in such a way that their main magnetic field more positive results. Adjust the position of either transformer until minimum hum is heard in the headphones. In actual radio receivers you can, of course, adjust the position of the transformer until minimum hum is heard.

Twisting of Power Supply Leads

Having seen how the low-frequency magnetic fields which are produced by power transformers and power pack filter chokes can be prevented from straying, we are ready to consider how the connecting leads to these parts can be prevented from setting up interfering magnetic fields. The power line leads which pass through the OFF-ON power switch to the power transformer are one possible source of trouble, and the low voltage, high current A.C. leads to the filaments of all tubes are another common cause of trouble. A study of the magnetic fields produced when two wires are close together will show how these magnetic fields can be suppressed.

Figure 16A shows two parallel wires, W_1 and W_2 . Alternating current flows out over one wire and back to its source over the other, but at any instant of



time the electron flow is in opposite directions as indicated. The magnetic lines of force produced by these currents have circular loops as indicated, with loops x produced by W_1 and loops y produced by W_2 .

Some of the magnetic loops will enclose both wires; notice that when this occurs, the magnetic lines of force will be in opposite directions and will cancel. Some of the magnetic loops will surround only their own wires, and in these cases the magnetic lines of force produced between the wires will all be in the same direction, and will reenforce each other.

Separating the wires allows more of these magnetic lines of force to pass between the wires and reenforce each other, and consequently the external 'magnetic field in the vicinity of the two wires becomes stronger. Placing the wires closer together, on the other hand, forces more of the magnetic lines of force to encircle both wires and cancel each other, thus reducing the strength of the external field. -Twin conducting leads should therefore be very close together if interfering magnetic fields are to be reduced to a minimum.

Even greater reduction in the external magnetic field produced by two current-carrying wires is possible when the wires are twisted together. The wires must, of course, be insulated in this case in order to prevent shorting them together. Figure 16B shows how twisting or transposing leads W_1 , and W_2 makes the individual flux paths xand y take opposite directions. In the vicinity of the wires these lines of force do not entirely interlock and cancel, but at a distance from the leads they actually do cancel and greatly reduce the external magnetic fields.

The more twists or transpositions there are in a given distance for two wires, the more reduction there is in external or stray magnetic fields. Incidentally, this reduction in the external magnetic field actually reduces the inductance of the conductors (any two wires which are close to each other and in the same circuit have a certain amount of inductance due to the fact that they form a single-turn coil).

Now you can understand why filament leads in radio apparatus are often twisted together or run very close together, and you can likewise understand why the power line leads to the power transformer primary are also twisted or placed close together. In some receivers, particularly those having low A.C. filament currents, manufacturers have found it unnecessary to twist filament wires or run them close together; elaborate precautions are needed only when trouble is actually encountered.

You will usually find that the power transformer is located at a point on the chassis very close to the place where the power line cord enters. The power switch, however, is usually located at the front of the chassis; the two leads running to it are usually quite long, and hence should be twisted together for minimum magnetic effects.

There is still a certain amount of leakage magnetic flux even when leads are twisted, and therefore filament and power supply leads should be kept as providing conductive paths for signal currents.

With the heater-type tubes which are so widely used in radio receivers at the present time, there is no direct connection between the filament and the cathode; if there is no leakage between these two elements, there will be no chance for signal currents to get to the filament circuit and for power line A.C. to get out of the filament circuit into signal circuits.

In the audio stages of radio receivers (particularly the power stages) and in the R.F. stages of radio transmitters,



FIG. 17. When a filament type tube is being supplied with an A.C. filament voltage, as at A, the method shown at B, C or D is used to prevent the A.C. filament voltage from affecting the grid or plate circuits.

far as possible from the grid leads in the first stages of a receiver or amplifier. Even extremely weak interfering signals entering these stages may become appreciably strong after amplification by the entire system.

Center-Tapped Filament Connections

The importance of preventing alternating current in filament leads from affecting signal circuits by magnetic induction has already been pointed out; it is just as important to prevent alternating current in the filament itself from affecting the signal circuits and to prevent filament leads from filament type tubes are still widely used; here the filament serves also as cathode, and hence preventive measures must be taken.

Let us first see what happens when we connect a filament-type tube as indicated in Fig. 17A. Consider a simple circuit like that in Fig. 17A, where a C bias battery provides the normal negative bias voltage for the grid of the tube and the filament is fed with an A.C. voltage of, let us say, 10 volts. The grid and plate return leads for signal currents go to point a on one filament lead. Point 2, at the electrical center of the tube filament, can be considered as the cathode of the tube, even though all points on the filament are emitting electrons. (For each point on the filament which is more negative than point 2 at any instant, there will be another point which is an equal amount more positive than point 2, regardless of the direction of current flow; as a result, the average of the voltage values between the grid and each point on the filament will equal the voltage between the grid and point 2, and likewise the average of all filament-to-plate voltages will equal the voltage between the plate and point 2.) The plate current will, therefore, be determined by the potential between point 2 and the grid (the grid bias voltage) and by the potential between point 2 and the plate (the D.C. plate voltage).

We can readily see, now, that the effective grid voltage will be the C battery voltage plus the voltage drop between points 1 and 2 on the filament. With 10 volts A.C. applied between points 1 and 3, there will be a 5 volt A.C. drop between points 1 and 2. This voltage drop will act in series with the C battery voltage insofar as the grid is concerned, making the grid bias voltage alternately 5 volts greater and 5 volts less than the battery voltage. Naturally this A.C. voltage of 5 volts applied to the grid will cause the plate current to vary correspondingly. The A.C. voltage drop will likewise act in series with the B battery voltage in determining the effective plate voltage, but the variations in voltage will obviously be more serious in the grid circuit because of the high amplification of the tube. Clearly the arrangement in Fig. 17A will result in excessive A.C. hum interference.

A theoretical solution to this problem is indicated in Fig. 17B, where the A.C. filament voltage is eliminated from the signal circuits simply by connecting the plate and grid return leads directly to point 2 by means of a tap on the filament. Since few if any tube filaments are made with center-tap connections, this solution cannot be utilized in practice.

The practical solutions to this problem, indicated in Figs. 17C and 17D, involve the use of external electrical mid-taps between the two filament leads. These mid-taps can be provided by a center-tapped resistor, as in Fig. 17C, or by a center-tapped filament winding, as in Fig. 17D. Often the midtap on the resistor is made variable, so it can be adjusted to compensate for unbalanced filaments and other conditions. When filament-type tubes are used in radio frequency amplifiers, bypass condensers are generally connected between the mid-tap and each filament lead to provide low-reactance paths to either filament terminal of the tube for signal currents; in the case of Fig. 17D, these condensers also serve to prevent signal currents from taking the path through the filament winding of the power transformer.

Analysis of Signal Current Circuits in a Typical Radio Receiver

The complete schematic circuit diagram of a typical 5-tube superheterodyne receiver is shown in Fig. 18 just as you might find it in the service manual for the set. We will study this diagram in detail, first tracing the paths of signal currents, then locating and identifying the various extra parts used by the designer to keep signal currents in their correct paths. Having done this, we will analyze the chassis layout of the receiver to see how the effects of stray electric and magnetic fields are minimized by proper placement of parts. This analysis will not only serve to summarize for you the

many practical facts covered in this lesson, but will also give you additional experience in reading circuit diagrams.

A practical radio man is able to read a circuit diagram like that in Fig. 18 almost at a glance simply because he has learned to consider only the important fundamental signal circuit parts, neglecting the various resistors and condensers which keep signal curfrom left to right. The R.F. signals picked up by the antenna are transferred to a grid of the first tube by T_1 , which we recognize as a tuned-secondary antenna transformer. The first tube has five grids and is therefore a pentagrid converter serving as mixer, first detector and oscillator. The first two grids of the tube act with parts C_3 , L_3 and L_4 as an oscillating circuit



FIG. 18. Complete schematic circuit diagram of a representative 5-tube superheterodyne receiver. This circuit is presented here only to show circuit functions and illustrate the filtering and by-passing methods described in this book; it is not intended for constructional purposes, and does not represent any one particular manufactured receiver.

rents in their correct paths, and neglecting the power supply leads. In Fig. 19 are shown only those parts which serve to identify the various stages of this receiver circuit; we will study this at first, in preference to the complete diagram in Fig. 18, for you are not as yet expected to be able to eliminate the less-important parts mentally.

We start at the upper left in Fig. 19, since the upper part of a radio circuit diagram is invariably arranged to read which produces the desired local R.F. carrier to mix with the incoming R.F. signal and produce in the plate circuit of this 6A7 tube the desired I.F. signal.

Adjustable condensers C_6 and C_8 identify T_3 as an I.F. transformer which passes on this I. F. signal to the 6D6 first I.F. tube for amplification, while I.F. transformer T_4 transfers the amplified I.F. signal to the diode section of the type 75 tube. Here you will recognize the conventional second detector-A.V.C. arrangement, with diode current developing across the detector load the desired A.F. component and a D.C. component of voltage for A.V.C. purposes. The first detector and the I.F. amplifier are A.V.C.-controlled for their control grid leads do not_return directly to their respective cathodes.

The A.F. signal is fed through coupling condenser C_{17} to the grid of the

through output transformer T_5 .

D.C. voltages for the various tube electrodes are supplied by the power pack, made up of power transformer T_6 , a type 80 full-wave rectifier tube, a filter system containing C_{24} , filter choke CH, C_{25} , the loudspeaker field (which serves also as a filter choke) and C_{26} . Voltage divider resistors R_{16} ,



FIG. 19. Simplified version of the 5-tube superheterodyne receiver circuit shown in Fig. 18.

triode section of the 75 tube, which consequently acts as an A.F. amplifier stage. The value of the A.F. voltage which is fed to this first A.F. stage is controlled by the setting of manual volume control potentiometer R_8 . The amplified A.F. signal is transferred to the type 42 output tube through resistance—capacitance coupling, and the A.F. output of this tube is fed to the voice coil of the dynamic loudspeaker R_{17} , R_{18} and R_{19} then provide the required values of B and C voltages, with C_{27} providing additional filtering for the C voltages.

Now that we are familiar with the general function of each stage in our receiver circuit, we can return to Fig. 18 and try our skill at identifying the various by-pass condensers and signal current filters which must be added to the basic circuit of Fig. 19 in order to

secure satisfactory performance. We see immediately that C_1 and R_1 act as the A.V.C. filter for the first detector, for R_1 connects to a point in the load circuit of the diode second detector. C_1 also serves to provide a path to ground and cathode for grid circuit R.F. signals, and to prevent shorting to ground of the D.C. A.V.C. voltage.

Next we come to the R_2 - C_4 combination. Since R_2 connects to the C point of the power pack voltage divider, we know this is not an A.V.C. filter; it is therefore a simple resistor-condenser signal current filter, which keeps oscillator R.F. signal out of the power pack. C_4 also permits application of the C bias voltage to the grid without interrupting the continuity of the oscillator tank circuit C_8 - L_8 for R.F. signals.

 R_3-C_5 is likewise a signal current filter, serving to keep screen grid (grids 3 and 5) and oscillator plate (grid 2) R.F. currents out of the power pack and at the same time by-pass these signal currents to the cathode of the 6A7 tube.

 R_4 - C_7 is a signal current filter which by-passes plate signal currents of the 6A7 tube to its cathode and at the same time keeps these currents out of the power pack.

 R_5 - C_9 are readily recognized as another A.V.C. filter once you trace connections from R_5 to the second detector.

 C_{10} is clearly the screen grid by-pass condenser for the first I.F. amplifier.

 R_6 - C_{12} form a signal current filter which by-passes plate signal currents of the first I.F. tube to cathode and keeps them out of the power pack.

Dual-condenser filter combination C_{14} - R_7 - C_{15} serves to keep R.F. current out of diode detector load R_8 by providing low-reactance paths to cathode for this current. C_{16} and R_9 together

serve as a signal current filter which permits application of normal C bias (developed across R_{17}) to the 6A7 and 6D6 A.V.C.-controlled tubes while preventing power pack hum currents from reaching these tubes and at the same time keeping signal currents out of the power pack.

 C_{17} is simply an A.F. coupling condenser and D.C. blocking condenser. R_{11} and C_{18} act as a signal current filter in providing a return path to cathode for A.F. grid currents of the first A.F. amplifier, keeping these currents out of the C bias supply circuit, and also serve to keep power pack ripple currents out of signal circuits. R_{10} provides a conductive path for the D.C. bias voltage, since it is high in ohmic value, the entire A.F. voltage fed through C_{17} by potentiometer R_8 is developed across it.

 C_{19} is an R.F. by-pass condenser which shunts to ground any R.F. currents which may reach the plate of the type 75 tube. R_{12} is the load resistor for this tube, with signal current filter R_{13} - C_{20} providing a path to ground and the cathode of the 75 tube for A.F. currents after they leave the load, thereby keeping these currents out of the power pack; R_{13} and C_{20} also prevent power pack ripple currents from entering the plate circuit.

 C_{21} is a D.C. blocking and A.F. coupling condenser, while R_{15} and C_{22} act as a signal current filter for the grid of the type 42 output tube. R_{14} is a grid resistor which permits application of the bias voltage to the grid.

Filament current for the four signal circuit tubes is provided by a separate winding (marked XX) on the power transformer. Since heater-type tubes are used, there is no electrical connection between filament and signal circuits, and no filter or by-pass condensers are required. The filament leads will, of course, be twisted or run close together under the chassis to lessen their magnetic effects on adjacent wires and parts. Notice the dotted line in the core of power transformer T_6 ; this represents an electrostatic shield made of a layer of metal foil actually placed between the primary and secondary windings to eliminate electrostatic coupling which might, under certain conditions, transfer undesirthat just studied is given in Fig. 20, which represents the top view of the chassis. In a receiver of this type, only the parts shown in Fig. 20 are mounted above the chassis; all the other parts are located underneath, with the chassis serving as an effective shield. Notice how parts which are connected together are placed close together; for example, the oscillator tuning condenser is alongside the oscillator coil, and the antenna transformer and tuning



FIG. 20. Logical chassis layout (top view) for the 5-tube superheterodyne receiver circuit of Fig. 18.

able power line disturbances to the secondary winding and the receiver circuits.

We have thus made a thorough analysis of the receiver circuit. Very seldom does a practical radio man have to study a circuit as thoroughly as this, but since he may have to analyze a particular section of the circuit in which trouble has developed, it is well for you to go over complete circuits like this occasionally during your training period.

A common and widely-used arrangement of parts for a 5-tube receiver like condenser are likewise near each other. The signal circuit tubes and transformers are arranged in the same order as on the circuit diagram, as indicated by the arrow line in Fig. 20. In any radio receiver you will find the parts so arranged on the chassis that the path taken by signal currents from tube to tube does not cross itself anywhere, with the end of the path (the output circuit) as far as possible from the starting point (the antenna circuit); this lessens the chances for feed-back and simplifies the shielding problem.

Notice also that the power pack com-

ponents-the power transformer, filter choke, filter condensers and the type 80 rectifier tube-are all grouped at one end of the chassis, as far as possible from the antenna input circuit, with the power cord entering the chassis just behind the rectifier tube. This lessens the chances for A.C. power line hum to get into the input circuit and be amplified by the entire receiver along with signal currents. You will find that the power transformer is completely shielded by its steel housing. As a further precaution against magnetic interaction, the filter choke is located at right angles to the power transformer.

All signal circuit coils and transformers above the chassis are in cop-

Be sure to number your Answer Sheet 20FR-3.

Place your Student Number on every Answer Sheet.

Never hold up one set of lesson answers until you have another ready to send in. Send each lesson in by itself before you start on the next lesson. In this way we will be able to work together much more closely, you'll get more out of your Course, and you will receive the best possible lesson service.

- 1. In what three different ways can undesired signal currents enter a circuit?
- 2. Why is a by-pass condenser commonly placed across the C bias resistor in the cathode circuit?
- 3. How can signal currents be kept out of the high-resistance path through the plate supply of a radio 10. circuit?
- 4. Why do signal currents take the path through a by-pass condenser in preference to other possible paths between two points in a receiver circuit?

per or aluminum shields, which prevent interference effects of stray electric and magnetic fields. Since the tubes all have glass envelopes, they are likewise covered with metal shields.

Wiring and leads for all components under the chassis of this receiver will in general be as short as possible, with all chassis connections for any one stage being made to the same point to eliminate the effects of stray signal current in the chassis. If you encounter leads in a manufactured receiver which seem unnecessarily long, do not change them unless you know definitely that they are causing trouble; oftentimes it is necessary to use long leads in order to get around certain parts and eliminate interfering effects.

TEST QUESTIONS

- 5. Can a signal current filter act both ways, preventing signal currents from getting out of a signal circuit and preventing A.C. power supply currents from getting into the signal circuit?
- 6. Why should all by-pass condensers in a single R.F. stage be connected to a common point in the stage?
- 7. Why must grid and plate leads in an R.F. stage be kept as short and as far apart as possible?
- 8. Is it important to ground an electric shield?

Name two materials which serve best as magnetic shields at high frequencies (above about 50 kc.)?

Why are chassis parts so arranged that the signal current path from tube to tube does not cross itself at any point, and its end (the output tube) is as far as possible from the starting point (the antenna circuit)?

9.

YOUR HEALTH

There is a definite relationship between a man's mental ability and his physical condition; for example, overeating is generally followed by a lazy feeling and a desire to sleep. The mind becomes less active. A headache may develop, along with a gloomy, crabby, or disgusted-with-life-in-general feeling. Certainly a man cannot do his best work when feeling this way.

Blue Mondays are quite real, and are caused by too much food and too little mental and physical exercise on Sunday, combined with a troubled sleep or too little sleep Sunday night. It takes several days for the human system to get back to normal after a week-end of excesses, so it may not be until Wednesday that you work with a clear mind. Then you find it easy to concentrate, and work becomes a pleasure. You say to yourself: "Ilow much happier I would be if every day were like this!"

But every day can be like this—if you take proper care of yourself, with physical exercise each day in the open air, and a good sound sleep each night. See a doctor if your sleep is not entirely restful.

Give your health the attention it deserves, and you will be rewarded many times by increased happiness and increased success in your work.

J. E. SMITH



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STUDY SCHEDULE No. 21

For each study step, read the assigned pages first at your usual speed. Reread slowly one or more times. Finish with one quick reading to fix the important facts firmly in your mind, then answer the Lesson Questions for that step. Study each other step in this same way.

I. Practical Importance; Coil and Condenser as Oscillator; How a Vacuum Tube Oscillator Circuit Works ... Pages 1-8

This first section is extremely important, because it gives you the basic principles upon which practically all radio oscillator circuits depend. If you understand these principles, you will not have to bother remembering the details of the many variations in oscillator circuits which are encountered in radio. Answer Lesson Questions 1, 2, 3, 4 and 5.

2. Effects of Circuit Changes; Type of Self-Excited

If you study this section with the idea that someday yon will actually be working with oscillator circuits and dealing with the various problems taken up, you will find this section intensely interesting. Try to locate the important idea in each paragraph, as an aid to understanding important facts. Answer Lesson Question 10.

Now we come to special types of oscillator circuits which, though interesting, need not be fully mastered at this time. If you get the general idea of how each of these circuits works, so that you can recognize the circuit when you encounter it, you can always return to this section later when you actually are working on these types of oscillators. Nevertheless, this section as well as every other section in this lesson will require careful study, because the subject of oscillators is so important to a radio man.

5. Mail your Answers for this Lesson to N.R.I. for Grading.

6. Start Studying the Next Lesson.

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The Vacuum Tube as an A. C. Generator in Radio-Television Circuits

PRACTICAL IMPORTANCE OF TUBE OSCILLATORS

RADIO, as a means of transmitting intelligence through space, depends upon the production of high frequency A.C. currents, and television requires the production of ultra high frequency currents. The operation, efficiency and stability of the super-heterodyne type of receiver depends upon the oscillator, a generator of A.C. currents. Some of the most successful electronic control systems are possible only because of the tube oscillator. Maintenance, testing and servicing of radio equipment can be satisfactorily carried out only with special oscillators called signal generators. Thus the very existence of sound and television broadcasting depends upon tube oscillators.

THE COIL AND CONDENSER AS AN OSCILLATOR

Long before radio was even considered as a public servant, scientists knew that an A.C. current would flow through a coil when it was connected to a charged condenser. The frequency of this current, as you already



VIBRATING ARMATURE

FIG. 1A. An early oscillator circuit; each closing of the buzzer contact recharges or primes condenser C in the coll-condenser circuit, and oscillations continue in the L-C tank eircuit when the buzzer contact opens.



know, essentially depends upon the inductance and capacitance values. The duration of the current is fixed by the resistance in the circuit; as the energy travels back and forth between coil and condenser the losses in these two parts reduce the energy a little on each "trip," until finally the oscillations cease. Clearly it is necessary to have some means of "priming" the circuit, or feeding it with at least enough energy to overcome resistance losses; the first scheme utilized a buzzer which connected the oscillatory circuit intermittently to a battery, as in Fig. 1A, so it would recharge the condenser regularly.

Spark Transmitter. In order to get greater power output, radio pioneers developed the high voltage oscillatory circuit of Fig. 1B; a step-up transformer having a vibrator in series with its primary (the combination being

called an *induction coil*) primes the oscillating coil-condenser circuit hundreds of times each second. The frequency of R.F. oscillation is determined essentially by the values of L and C.* Of course there were many later improvements on this early spark transmitter circuit, but these are now "ancient radio history," mentioned simply to show that the coil and condenser were first and still are the important frequency controlling units.

Damped Waves. Each time the condenser in an oscillating circuit is charged, the applied energy immediately starts to oscillate between condenser and coil; the resistance of the oscillatory circuit (essentially the resistance of the coil or of coil and spark gap together) causes the amplitudes of successive cycles to be reduced or damped, as in Fig. 2A. Repeated priming of the oscillatory circuit (at points 1, 2 and 3) give a series of such oscillation groups. Although this form of oscillation serves very well for code transmission, it has the very severe drawback that it creates excessive interference with adjacent-channel stations, often "riding in" with signals of other stations. Reducing the resistance of the oscillatory circuit improves matters by giving the wave form illustrated in Fig. 2B, but still the damped oscillations are unsuited for the transmission of speech and music. The reason for this is simple; the damped wave form is in itself modulated at a frequency which depends upon the number of oscillatory groups per second, and this modulation note would interfere with the program being transmitted.

A continuous, constant-amplitude wave form like that shown in Fig. 2C, free of modulation, is required as a carrier when transmitting voice, music or picture signals. This wave can be produced by a vacuum tube oscillator, a simplified circuit of which appears in Fig. 3.

Vacuum Tube Oscillators. In a vacuum tube oscillator the power loss due to the resistance in the L-C oscillatory circuit is constantly being compensated for by power which is supplied by the tube acting as an amplifier, whereas in the spark gap transmitter the losses were compensated for intermittently.

Let us see how a vacuum tube oscillator operates. Only electrical changes in the circuit, such as connecting the tube to its power supply, will start this vacuum tube oscillator; once oscillation starts, there is produced across the condenser (or coil) terminals an A.C. voltage whose frequency is determined by the values of L and C. The vacuum tube, acting as an amplifier, will respond to this varying grid voltage and cause the plate current to vary at the same frequency. Coil L_T , carrying this A.C. plate current, will induce into coil L by mutual induction a voltage e. Coil L, condenser C and resistor R (an apparent resistance, representing the combined resistances of L, C, the grid-cathode path in the tube, and the resistance effects of a load, if used) act as a series resonant circuit. The reactance effects of Land C cancel each other, and the voltage e acting on R determines the amplitude of the oscillatory A.C. current. As in any simple generator-load

k

[•]The frequency of an oscillatory L and O circuit can be determined from the following formula: $f = 159,000 + \sqrt{LO}$ where f = frequency of oscillation in cycles per second, L = coil inductance in microhenrys, and O = condenser capacity in microfarads.

circuit, the current assumes a value which will make the A.C. voltage drop in R equal to the induced A.C. voltage e.

Because of resonance, the voltage across either L or C is greater than the induced voltage; this resonance-amplified voltage, applied to the grid of the tube, drives the plate current through large swings, and the increased current through the tickler coil increases the induced voltage e. The plate current swing and consequently the induced voltage e continues to increase up to the limit of the amplifying ability of the tube; the tube thus introduces into the oscillatory circuit, through tickler coil $L_{\rm T}$, enough power to overcome power losses in R and make the oscillatory current assume the constant amplitude shown in Fig. 2C.

There are many ways of operating the oscillator tube, connecting the oscillatory circuit and producing the feed-back voltage in order to secure improved performance; these will now be studied. Remember, however, that it is essentially the values of coil L and condenser C in the oscillatory circuit which govern the frequency of a self-excited vacuum tube oscillator; these values can be adjusted to give frequencies ranging from one cycle to hundreds of kilocycles per second.

HOW A VACUUM TUBE OSCILLATOR CIRCUIT WORKS

Most tube oscillators work alike, even though the method of feed-back, the position of the oscillatory circuit containing C and L (also called the



FIG. 2. Priming or recharging of an oscillator circuit gives the series or "trains" of damped oscillations shown at A and B, but present-day radio transmission calls for the constant amplitude oscillation shown at C, which can be obtained from a vacuum tube oscillator. For simplicity, only a few cycles of the oscillatory wave are shown here; actually there may be thousands of cycles of oscillations in each damped wave group, such as between points 1 and 2 in A.

tank circuit), the method of supplying the D.C. power, or the positions of the circuit components may vary. A typical oscillator circuit is shown in Fig. 4; here, as in most oscillators, the plate feeds a part of its energy back to the grid to make the tube supply an A.C. plate current. All oscillators in which the grid takes a small part of the plate circuit power are said to be self-excited.

Power Input. Neglecting tube filament power, the total power fed to the oscillator is the D.C. plate voltage V multiplied by the D.C. plate current I. This input power is used in several ways: 1, It overcomes resistance losses in the L-C tank circuit. If coil L is coupled to a load, the effective value of R is increased and the power consumed by the tank circuit also includes useful output power or useful work. 2, The grid in an oscillator circuit is driven positive at times, causing grid current to flow, and the grid conse-

quently draws power from the plate circuit; 3, a part of the input plate power is used to overcome the A.C. plate resistance, thus heating the tube elements. In spite of all these power losses, from 50% to 85% of the input power can be converted into useful A.C. power in a good oscillator.

An oscillator circuit is unique in that changing any one thing in the circuit changes current and voltage conditions in the entire circuit; the reason for this is that the grid and plate circuits are linked together. A study of Fig. 4 will reveal a few outstanding facts which apply in general to all selfexcited oscillators. The A.C. voltage across C or L can never have a peak voltage greater than the applied voltage V; in fact, the peak value of $v_{\rm T}$ will always be less than V because the charging voltage of C is equal to the applied voltage minus the tube voltage drop. Furthermore, as the A.C. plate current *increases* to furnish more power to the tank circuit, the A.C. voltage across C or L becomes less.

A.C. voltage $v_{\rm T}$ causes an A.C. current $i_{\rm T}$ to circulate in the tank circuit, and this current induces an A.C. voltage in the grid circuit. The amount of voltage induced in the grid circuit will, of course, depend upon the mutual inductance between coils $L_{\rm G}$ and L, upon the A.C. tank current $i_{\rm T}$, and upon the frequency of oscillation. The mutual conductance of the oscillator tube must be such that the grid driving voltage causes enough A.C. plate current to flow to maintain oscillation; in other words, the mutual conductance $(g_{\rm m})$ of the tube is of as much importance as the A.C. grid voltage. Oscillator action is a sort of "around the circle" affair; the grid must draw enough power from the tank circuit to swing the plate current enough to produce enough A.C. power in the tank circuit to supply the load with power and feed sufficient power back to the grid to maintain oscillation. The amount of output power can be controlled by changing the mutual conductance of the tube or by changing the mutual inductance between the plate and grid circuits.

Starting an Oscillator. Getting oscillations to start in a feed-back circuit like that in Fig. 4 is often explained as follows: When the tube is first connected to the power supply, the plate current starts to rise as soon as the cathode begins to emit electrons. A rising plate current flowing through coil L (see Fig. 4) induces a voltage in coil L_0 . This grid voltage, acting through the mutual conductance of the tube, causes the plate current to vary; if the voltage to the grid is so phased that the original plate current change is aided, the plate current will build up to the full capacity of the tube, as determined by the operating voltages. This primes the tank circuit with energy and it starts to oscillate, feeding A.C. voltage to the grid and causing the plate current to increase and decrease alternately; this is the desired condition for a tube oscillator. Tank current and grid voltage continue to increase until a balance is reached, this occurring when the A.C. current in the tank circuit produces (through coupling to the grid) sufficient A.C. grid voltage for the tube amplifier to supply the tank power required by the tank circuit load and to overcome the grid and plate circuit power losses. All power is obtained from the plate supply source.

Automatic C Bias. Now let us concentrate our attention on grid resistor R_c and grid condenser C_c in Fig. 4. As you will shortly see, condenser

 $C_{\rm c}$ has a double purpose. The A.C. grid voltage $e_{\rm g}$ acts directly upon the grid-cathode of the tube because $C_{\rm c}$ has a low reactance. Thus the tube is fed with an A.C. signal which makes the grid alternately positive and negative with respect to the cathode. When the grid is positively charged, it will draw electrons from the cathode, these electrons flowing through resistor $R_{\rm c}$ on their way back to the cathode since they cannot pass through condenser $C_{\rm c}$; a D.C. voltage therefore appears across resistor $R_{\rm c}$.

From the basic fact that electrons flow through a load from the negative to the positive terminal, point 1 of resistor R_c is negative with respect to point 2. If condenser C_c were absent, the current flow through R_c would be of a pulsating nature since the grid is positive for only short periods of time; this condenser, however, is charged by the voltage drop across the resistor so that when the electron flow through R_c starts to decrease, condenser C_c begins feeding electrons into R_c . The result is a steady flow of electrons through R_c , and the voltage drop across it becomes constant. Thus, in addition to the A.C. voltage supplied by L_c , a steady D.C. bias voltage is automatically supplied to the grid by grid condenser C_c and grid resistor R_c ; this voltage is called the *automatic C bias*. This makes the operating plate current set itself at a lower value.





FIG. 3. Simple feed-back type vacuum tube oscillator circuit. R is not an actual resistor; it simply represents the resistance of coil L and all other resistances, apparent or otherwise, which may cause losses in the oscillatory circuit. Likewise e represents the voltage induced in coil L by the tickler coil.



You may consider the A.C. voltage and the automatic C bias voltage as acting independently on the tube, or you may consider both effects at the same time, as you prefer. The presence of the C bias lessens the effect of the A.C. grid voltage on the tube, actually reducing the time and the amount by which the grid is driven positive; the C bias cannot, however, completely prevent the grid from being positive, for if it could, there would be no grid current to produce the automatic C bias.

The automatic C bias can never be greater in value than the peak A.C. grid voltage. The average value of grid current multiplied by the ohmic resistance of R_c determines the C bias in the circuit. It should be pointed out that the grid bias resistor may be connected from grid to cathode (instead of across the condenser) without changing the automatic C bias.

Operating Efficiency. It is true that the circuit in Fig. 4 will oscillate without any form of C bias, but in this case the D.C. plate current will be so high that losses will be excessive, giving very poor tube efficiency at normal

plate voltage, and the resulting heat may melt the tube elements. The use of an automatic C bias keeps D.C. plate current at a minimum and limits the effects of the A.C. grid voltage, allowing the tube to work more efficiently. The value of R_c and the feed-back voltage are so selected that the average C bias at least equals the value for plate current cut-off (this C - bias approximately equals the plate voltage divided by μ , the amplification factor of the tube). This means that plate current flows for one-half ofeach cycle, and grid current flows during a much smaller part of the cycle.* The coil and condenser tank circuit stores up energy, and because of its oscillatory nature continues to oscillate during those parts of the cycle when there is no plate current; the resulting tube efficiency is at least 50%. By making the C bias even greater than cut-off value, the time of plate and grid current flow is reduced and greater efficiencies, which may be as high as 85%, are obtained. Although increasing the C bias does increase efficiency, it naturally reduces the power output; a compromise must therefore be made between maximum efficiency and maximum power, to suit a particular need.

A fixed C bias voltage (secured from a tap on the voltage divider of the D.C. power supply) could be introduced in place of R_c and C_c , but if it is made greater than the plate current cut-off value, the oscillator will not be self-starting (the original rise in plate current will not occur). Automatic C bias is a very practical solution, for the C bias generally sets itself automatically for best operation.

Blocking. The ohmic value of the grid resistor in a self-excited vacuum tube oscillator (such as R_c in Fig. 4) is not critical, but if it is made too large, its D.C. voltage will build up to such a high value that the oscillator will block (stop oscillating momentarily) at regular intervals. Values of R_c above about one megohm cause the oscillator to block, producing a tone modulation whose frequency depends upon the time constant of the resistor-condenser combination; the value of R_c in megohms multiplied by the capacity of C_c in microfarads gives the approximate time of one blocking and starting cycle. When the value of R_c is low (5,000 to 250,000 ohms), the bias is never driven so far negative that intermittent oscillation is obtained. The grid condenser C_c can be any size sufficiently large to make its A.C. reactance in the grid circuit negligible.

In a properly adjusted oscillator, the C bias is zero at the instant of starting the oscillator, and a definite value of plate current flows. This plate current decreases quickly to the operating value I_P as automatic C bias comes into action, for the operating bias is now more negative. This means that if the oscillator is stopped by shorting one of the coils (or by an internal short in tank condenser C), the plate current will *increase*; in some low power oscillators this is actually done to determine if the circuit is operating, but in high power oscillators the high plate current resulting from such a procedure might damage the tube.

^{*}These pulsating grid and plate currents produce many harmonics, but the tank circuit accepts only the fundamental. Coil L suppresses and condenser O by-passes the harmonics; by using a large value for C and a small value for L, the ability of the tank circuit to reject harmonics will be improved even more.

To adjust an oscillator to best operating conditions once the load to the oscillator is fixed, it is customary to adjust the different circuit parts to the values which give minimum D.C. plate current at the desired frequency. When an oscillator is to be tuned over a wide range of frequencies by varying the tank circuit capacity, as is the usual procedure, an optimum (best) circuit adjustment for one frequency may be unsatisfactory at some other frequency. For example, if the circuit is adjusted for best operation at a high frequency in its range, the resetting of the tank circuit to a low frequency may result in too low output or no oscillation due to too low a feedback voltage. When a wide range of frequencies is to be covered, it is best to make optimum adjustments at a mid-frequency; this should provide ample feed-back excitation at the lowest frequencies.



FIG. 5. These plate and grid currents and voltages represent operating conditions in the oscillator circuit of Fig. 4. Remember that graphs like these are always read from left to right. When comparing two voltages, that one which reaches a positive peak *closest* to the vertical reference line is said to *lead* the other; thus, pT in A leads eg in B.

Current and Voltage Relations. The curves in Fig. 5 present a picture of the grid and plate voltages and currents in the oscillator circuit of Fig. 4 during operating conditions; these curves are characteristic of all types of oscillators. Notice that although the applied plate voltage V in Fig. 4 is a D.C. voltage, the voltage $v_{\rm T}$ across the tank circuit (the plate-to-cathode A.C. voltage component) has essentially the sine wave form shown in Fig. 5A. Although the A.C. plate voltage swings above and below the D.C. plate supply voltage value, it never is driven to zero because of the A.C. plate resistance. In Fig. 5B is shown the A.C. grid voltage curve; you will notice that the A.C. grid voltage reaches its maximum positive value half a cycle (180 degrees) after the A.C. plate voltage reaches its maximum, or in other words, the A.C. grid voltage lags 180 degrees behind the A.C. plate voltage; coils L_0 and L in Fig. 4 are so connected that this phase relation is produced. In this way the grid can be driven positive during the time when the plate-to-cathode voltage is low; current thus flows in the plate circuit when the grid swings positive, and the tank circuit absorbs or stores up energy. This energy is released because of the oscillatory action (also called the flywheel effect) of the tank circuit when plate current drops to zero, completing the A.C. cycle and producing the sinusoidal voltage wave shown in Fig. 5A.

Looking at Fig. 5C, note that plate current $i_{\rm P}$ begins to flow when the instantaneous A.C. grid voltage becomes more positive than the cut-off value of C bias. Grid current (Fig. δD) flows only when the A.C. grid voltage overcomes the effective C bias, and makes the grid positive with respect to its cathode. Thus plate current flows for a longer part of a cycle than grid current. Since the flow of grid current robs the plate circuit of current, a dip appears in the peak of the plate current curve; a dip will also appear in the grid current curve if a large amount of secondary emission takes place at the grid of the tube. The average value of plate current controls the amount of power taken from the supply; in the grid circuit the current through the grid resistor is averaged out by the smoothing condenser. By studying curves A, B and C together, you can see that the minimum plate voltage value is very important in controlling plate current; the larger this minimum voltage, the larger the plate current.

An Important Oscillator Requirement. Any self-excited oscillator must have the same phase relationship between grid and plate A.C. voltages that exist in an ordinary amplifier—the A.C. grid voltage must be 180 degrees out of phase with the A.C. plate voltage. For example, in an amplifier having a resistance load, a positive swing in grid voltage makes the plate current increase, increasing the IR drop in the load and therefore decreasing the plate-to-cathode voltage; the tank circuit of an oscillator is in effect a resistance load at resonance, and the same relation must therefore exist—grid voltage increase causes plate voltage decrease.

EFFECTS OF CIRCUIT CHANGES

Changes in the adjustments of various parts of an oscillator circuit will naturally affect the oscillatory condition; the following discussion of these changes will in general apply to all self-excited oscillators.

Static and Oscillating Plate Current. If the coupling between plate and grid in an oscillator is reduced or the grid excitation supply is shorted, the circuit will not oscillate, and the plate current will assume a static value which is governed essentially by the plate supply voltage and the D.C. plate resistance (the grid bias being zero for a self-excited circuit which is not oscillating). If the circuit was originally adjusted for optimum efficiency as an oscillator, an unusually high plate current will flow; in high power oscillators the tube will overheat and plate and filament will be destroyed by heat. In the case of low power oscillators such as those found

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in testing equipment and in superheterodyne receivers, the excessive plate current may weaken the cathode emission qualities of the tube. When the fault is remedied and oscillation starts again, the D.C. plate current will immediately drop. In the case of high powered oscillators, the plate current should immediately assume its normal value (much lower than the estimated static value); if plate current is above normal the circuit should be disconnected from the supply at once.

Effect of Plate-to-Grid Coupling. There can, of course, be no oscillation when plate-to-grid coupling is very loose; as coupling is gradually increased, oscillation will start at a definite point and the D.C. plate current will drop from the static value to the operating value. Further increases in coupling simply increase the grid excitation (increase the grid voltage swing), thus increasing the rectified grid current value and shifting the operating grid voltage more negative, but D.C. plate current remains practically constant for all values of coupling which maintain oscillation. When coupling is increased beyond a certain critical point, the time duration of the plate current pulses becomes such a small fraction of a cycle that the tank circuit can no longer supply sufficient energy to build up the voltage wave for the rest of the cycle, and oscillation stops; plate current then rises to the static value. Any change in coupling also changes the inductance and resistance of the tank circuit, thereby changing the frequency of oscillation slightly.

Effect of Loading the Tank Circuit. The tube circuit of an oscillator must supply a certain amount of power to the tank circuit in order to maintain oscillation; this power comes from the D.C. supply, for there is no other source of power in a self-excited oscillator. If a load is coupled (usually inductively) to the tank circuit, the tube must also supply the extra power demanded by the load, and this extra power must likewise come from the D.C. supply. This means that the D.C. plate current increases in value when a load is applied to the tank circuit of a self-excited oscillator (since power is equal to voltage multiplied by current, and the D.C. supply voltage is here essentially constant).

The A.C. plate current must increase either in amplitude or in the duration of a pulse in order to produce the increase in average or D.C. plate current; actually both factors increase, for the reduced tank current lowers the grid excitation, making the C bias more positive and allowing a current pulse of greater amplitude to flow for a slightly greater fraction of a cycle.

Effect of Plate Supply Voltage. Increasing the D.C. plate supply voltage of a self-excited vacuum tube oscillator increases the A.C. tank voltage, thereby raising the tank current and *increasing the power output*. The higher tank current also serves to increase the grid excitation, the rectified grid current, the automatic C bias and to a slight extent the D.C. plate current. If a self-excited oscillator will not start by itself, increasing the plate supply voltage will generally cure the trouble.

Effect of Increasing C Bias Resistor. Increasing the ohmic value of the C bias resistor makes the grid bias voltage more negative, with the result that grid and plate currents flow for smaller fractions of the cycle. Other effects are a lowering of the D.C. plate current, a decrease in the amount

of energy supplied to the tank circuit and an increase in oscillator efficiency. These changes are not very great, for the increased efficiency tends to feed more grid A.C. voltage to the oscillator input, offsetting the increased bias. If the resistor value is increased too much, repeated blocking will take place, and stability (the tendency to remain in oscillation) will be reduced:

General Circuit Effects. It is desirable to have a high inductance-toresistance ratio (high Q factor^{*}) in the tank circuit of an oscillator in order to improve the flywheel action of the circuit and reduce the generation of harmonics in the tank circuit. On the other hand, a high Q factor gives a high tank current, increasing circuit losses unnecessarily. Circuits which are heavily loaded have a low Q factor and are very unstable, hence a compromise involving a mid-value of Q factor is usually required.

Changing the capacity in the oscillatory circuit naturally changes the frequency of oscillation; as the condenser capacity is reduced, the frequency goes up. The feed-back voltage or grid excitation depends upon the coupling (usually fixed), upon the tank current and upon frequency, hence tuning an oscillator to a higher frequency increases the feed-back voltage. Although the losses in the tank circuit increase with frequency, causing the tank current to reduce, this is more than overcome by increased feed-back. Increased feed-back makes the oscillator generate more power, but this effect is somewhat reduced by the resulting increased grid current and increased C bias voltage. By limiting feed-back, it is possible to secure an oscillator which can be tuned over a wide range of frequencies.

On the other hand, an oscillator set for optimum conditions at a high frequency may stop oscillating at the lower frequencies because of insufficient feed-back. In superheterodynes an oscillator may fail at lower frequencies because of poor tube emission, low plate voltage, or too high a C bias resistor value. Remedies include trying a new tube at normal voltage or adjusting the plate voltage and C bias resistor values to give oscillation at the lowest frequency desired with a given tank circuit coil and condenser.

TYPES OF SELF-EXCITED OSCILLATORS

As I pointed out before, the basic oscillator ideas which you have been studying apply in general to all types of self-excited oscillators. Now we will take up the common variations of the self-excited oscillator circuit.

Tuned Grid Oscillator. This is of the form shown in Fig. 6A. The tank circuit is in the grid circuit, plate A.C. power being fed to it by inductive coupling. The circuit as shown here is for low power oscillators, such as those found in superheterodynes and in test oscillators. To prevent over-excitation and to permit greater tank current, the grid is often connected to a tap on the tank coil; in this way the tank circuit can be inductively loaded, and the grid excitation can be adjusted to a satisfactory minimum value.

^{*}A parallel resonant circuit acts as a large resistance, even though the circuit parts themselves have comparatively low resistances. The number of times the resistance of a parallel resonant circuit has been increased by tuning is called the Q factor of the circuit; this Q factor is equal to $\omega L \rightarrow R$, where ω is 6.28 times the frequency in cycles, L is the inductance of the coil in henrys and R is the resistance of the circuit in ohms.



FIG. 6. Simplified schematic circuit diagrams of eleven different types of self-excited vacuum tube oscillators. Heavy lines indicate the coil and condenser tank circuit combination which controls the frequency of the oscillator. Although there are many possible variations of each of these circuits, you can identify a particular oscillator by the way in which its frequency-controlling tank eircuit is connected to the grid and plate electrodes of the tube. Condensers marked C are frequencycontrolling condensers; C₁ and C₂ are oscillation-controlling condensers; Cc is the low A.C. reactance condenser, which also smooths the rectified grid current; CP is a low reactance A.C. by-reacondenser; C is a D. C. blocking condenser. When a condenser is double-marked, as CB and CP, it acts as a blocking condenser for D.C. and as a low reactance condenser for A.C.

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Tuned Plate Oscillator. You are already quite familiar with this simple but effective oscillator circuit, shown in Fig. 6B and also in Fig. 4. The tank tuning condenser is in the plate circuit; changing the mutual inductance between grid and plate coils changes the grid excitation. This circuit is better than that in Fig. 6A for high frequency power generation. Often the plate is connected to a tap on the tank coil to get a better impedance match between the tank circuit and the A.C. plate resistance of the ______ tube;* this gives greater efficiency.

Meissner Oscillator. This circuit, named after its inventor, is shown in Fig. 6C; the fact that feed-back and grid excitation are essentially independent of each other is an outstanding feature. Though not intended for high power requirements, the circuit is easily adjusted for optimum conditions, and gives highly stable operation. The tank circuit is usually coupled loosely to the grid and plate coils, giving the tank coil and condenser practically perfect control of the oscillating frequency. The isolation of the tank circuit also permits grounding of the condenser without affecting the supply circuits or R.F. circulating currents.

Hartley Oscillator. Also named after its inventor, the Hartley is perhaps the most widely used oscillator in the radio field. The basic circuit is given in Fig. 6D. The Hartley oscillator can always be identified by the fact that the tank circuit is in both the grid and plate circuits, the division of voltages being made by a tap on the tank coil. This tank coil tap, which connects through an R.F. by-pass condenser to cathode and ground, divides the relative plate and grid R.F. voltages without disturbing the frequency of oscillation. Moving the tap away from the plate reduces the grid excitation. The optimum condition, when D.C. plate current is at a minimum value, is easily obtained. Additional improvement in efficiency is often realized by connecting the plate to a movable tap on the tank coil, especially when low plate impedance tubes are used.

Colpitts Oscillator. Where highly stable oscillators are needed, this circuit (shown in Fig. 6E and named after its originator) is widely used. A unique feature is the use of two tuning condensers to divide the R.F. voltages between plate and grid circuits; individually the condensers determine the plate and grid A.C. voltages, but together in series they determine the frequency of oscillation. Decreasing the grid to cathode capacity increases the grid excitation, but the plate to cathode capacity must then be increased to maintain the frequency of oscillation. These adjustments are usually carried out until minimum D.C. plate current is obtained. A lowcapacity variable condenser is often connected in parallel with the tank coil to make the final frequency adjustment. Because the division of A.C. plate and grid voltages is independent of frequency, the inductance can be changed from one frequency band to another without affecting the adjustment for best operating conditions. In a few of the low powered Colpitts oscillators the correct distribution of plate and grid capacity is determined beforehand and the two units ganged (their rotors mounted on the same shaft) for quick variation of frequency in a given band.

^{*}With this connection the peak amplitude of the tank voltage will be greater than the applied D.C. plate voltage.

Ultra-Audion Oscillator. As one of the oldest of oscillator circuits, this seems to have been used long before the reason why it worked was known. The tank circuit is connected between grid and plate, the D.C. plate current being kept out of the grid by a blocking (grid) condenser, as shown in Fig. 6F. This circuit oscillates because the plate-to-cathode and grid-to-cathode tube capacities (indicated as C_{PK} and C_{GK}) form an A.C. voltage divider which makes the circuit a variation of the Colpitts oscillator. At very high R.F. values the tank tuning condenser may be eliminated, the tube capacities serving both for voltage division and for tank tuning. The circuit may be adjusted for the optimum operating condition by varying the plate voltage and the grid resistor value. A midget type variable condenser is often connected between the grid end of the tank circuit and the cathode, to give control of the intensity of oscillation.

Tuned Grid, Inductive Plate Oscillator. This circuit, shown in Fig. 6G, is on the whole an undesirable (parasitic) oscillator, more often eliminated from than introduced into a system; it is one of the reasons for oscillations in amplifying stages. The operation of the circuit is such that when the plate load is an inductance, the A.C. voltage which is fed back to the grid circuit through the grid-to-plate capacity of the tube is in phase with the A.C. grid voltage. This feed-back voltage thus aids the A.C. grid voltage, resulting in large sustained plate current swings and a transfer of power from the plate circuit to the grid circuit. If the plate inductance is large (high reactance), sufficient voltage will be fed back to overcome grid circuit losses and sustain oscillation. The frequency is, of course, controlled by the oscillatory circuit connected to the grid. The oscillations can be reduced by shunting the plate coil with a resistor, by introducing a resistor into the grid lead, by reducing the plate voltage, by introducing plate-to grid capacity-bucking components (so called neutralizing circuits), and by using screen grid tubes. However, the tuned grid, inductive plate oscillator does serve a highly useful purpose, when the tank circuit is replaced with a crystal oscillator.

Tuned Grid, Tuned Plate Armstrong Oscillator. The so-called Armstrong (the inventor) tuned grid, tuned plate oscillator is shown in Fig. 6H. This circuit is essentially like that shown in Fig. 6G, except that a parallel resonant plate circuit is used to create the inductive plate load which is necessary for oscillation. When a parallel resonant circuit is tuned slightly above the frequency of the R.F. source, its impedance becomes highly inductive. In this circuit the grid is tuned to the desired frequency and the plate tuned for minimum D.C. plate current, which automatically sets the plate tank circuit at a high inductive value. The grid tank circuit is often replaced with a crystal oscillator and the plate circuit adjusted approximately to minimum plate current; this gives a widely used and very important crystal circuit, to which we will return shortly.

Tuned Plate, Cross-Fed, Push-Push Oscillator. In Fig. 61 you have a circuit which is often used where both high efficiency and stability are desirable. Although a double triode tube is shown, two similar triode tubes are more often used. Each tube is automatically biased to cut-off value and each grid is fed 180 degrees out of phase with the other, so that when one tube is drawing plate current, the other tube is idle. In this way the

tank circuit, which is connected directly to the two plates, is receiving energy for both halves of the cycle. If during one-half of the cycle point 1 is positive with respect to point 2, then point 2 will be positive with respect to point 1 for the next half cycle. Point 1 being coupled to the grid of tube VT_2 through a small variable condenser C_2 , this tube receives an A.C. grid voltage which is *in phase with* its normal excitation. Point 3 and the cathode are at the same R.F. potential. Another way of looking at this circuit is to remove one tube; now we have in effect an ultra-audion circuit. Increasing the capacity of feed-back condensers C_1 and C_2 increases grid excitation; both condensers are usually adjusted for minimum D.C. plate current. This circuit is widely used at high and ultra-high frequencies, because the tank circuit is shunted by only one-half the plate-to-cathode capacity; this allows operation at higher frequencies without encountering undesirable tube capacity shunting effects.

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Tuned Grid, Tuned Plate Push-Push Oscillator. Here is another favorite dual tube circuit, relying on grid-to-plate tube capacity for feed-back; its circuit appears in Fig. 6J. This circuit is essentially like the Armstrong circuit, except that the plate tank circuit is being fed with energy on both halves of the A.C. cycle, giving greater efficiency. Like all twin tube oscillators, it is a valuable circuit at high frequencies where tube capacities become important. The circuit frequency depends upon the grid tank circuit; the plate tank circuit is adjusted for minimum plate D.C. current, which sets the plate load to an inductive condition.

TNT Push-Push Oscillator. This circuit, shown in Fig. 6K, uses a coil alone in the grid circuit, relying on distribution coil capacity for grid resonance; the frequency of oscillation is therefore limited to a narrow band. This gives a high L/C ratio for the tuning circuit in the grid, which is excited to large A.C. voltage values by the limited tube capacity feed-back. The plate tank circuit is tuned to an inductive value.

Direction of Plate and Grid Coil Windings. When the plate is inductively coupled to the grid, the feed-back voltage may be either in or out of phase. If out of phase, the oscillator will not work; in this case, reversing connections to one of the coils will make the oscillator start. A good rule to follow in connecting oscillator coils (assuming that the two coils are wound in the same direction) is to make the two outer coil terminals or leads (1 and 4 in Fig. 6A) the cathode and B+ terminals, leaving the grid and plate connections for the inner two coil ends, 2 and 3.

APPLYING OSCILLATOR LOADS

Oscillator loads must be applied in such a manner that load changes have a minimum effect on frequency change, and do not destroy the stability of the oscillator. As you know, the tank circuit current and voltage are sinusoidal in wave form, and therefore are of the fundamental oscillator frequency. To feed the fundamental oscillator signal to the load, then, it is best to couple the load to a *tank circuit inductance*; several different methods are used.

Inductive Coupling to Load. This method, shown in Fig. 7A, is very commonly used when the effective load resistance is low (is 500 ohms or

less). The effect of inductive coupling is that of a resistance in series with the tank coil and condenser, the resistance increasing in value as the mutual inductance is increased. Of course, this method of coupling affects the resonance of the tank circuit, shifting the frequency of oscillation slightly when load is applied or varied.

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Capacity Coupling to Load. To be sure, high resistance loads can be coupled inductively to the oscillator if a large mutual inductance is provided; in cases like this, however, the capacity coupling method shown in Fig. 7B is more often employed. The effect of capacity coupling is that of a high resistance placed in parallel with all or a part of the tank circuit. Any parallel resonant circuit acts as a pure resistance at resonance; connecting a resistance load in parallel merely decreases the net value of resistance. This loading may be increased (by reducing the resistance of the load) nearly to the point where the circuit will no longer maintain its selfexcitation. The electrical value of coupling condenser C is such that its reactance is negligible with respect to the reactance of the load, and hence C does not affect the loading on the oscillator. The loading can be increased very effectively, in cases where the value of the load is fixed, by moving the load tap on the tank coil away from the R.F. ground.





After an oscillator is loaded, it must be adjusted for optimum operation by changing the grid bias resistor value and by adjusting the grid excitation until minimum D.C. plate current is obtained, making sure that the expected power output is obtained at all times. Where the effects of the load are to be minimized, as when the oscillator is being tuned over a given band, a series resistor of high value is placed at point X in Fig. 7B, or the inductive coupling to the load is reduced. Of course, this cuts down the available output power, but stability is far more important than high power in the superheterodyne receiver and the test oscillator, where capacity coupling is often used.

Electron Coupling to Load. Practical electron coupled oscillator circuits like that shown in Fig. 7C were first introduced by Dow, and hence are sometimes called *Dow oscillators*. These circuits are readily identified by their use of a multi-element tube in which the cathode, first and second grids are the basic oscillator elements (acting as the oscillator cathode. control grid and anode); these oscillator elements produce, just beyond the second grid, a pulsating space charge which acts as a virtual cathode. **The plate of the tube**, which is at a high positive potential and connects

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to the load through a coupling condenser, attracts electrons from the virtual cathode and passes them on to the load. The term "electron coupling" comes from the fact that a stream of electrons is the only coupling between the anode (second grid) of the oscillator circuit and the plate in the load circuit. The circuit in Fig. 7C is that of a Hartley oscillator with the second grid placed at an R.F. ground potential so it acts as a shield between the plate (or load) and the oscillator. Electron coupling gives a high stability with a minimum of load reaction on the oscillator. Although the Hartley oscillator is shown in Fig. 7C, any of the other oscillator circuits may just as well be used.

Best Position for Load. In loading one of the standard oscillator circuits, load that tank circuit which least affects the oscillator. In Figs. 6A to 6G, of course, the load is applied to the only tank circuit present; in Fig. 6G, however, the load might be coupled to the plate coil if this did not reduce the reactance of the coil too much. In Figs. 6H to 6K inclusive, the plate coil would be loaded, the plate tank being retuned for minimum D.C. plate current.

OSCILLATOR POWER SUPPLY CONNECTIONS

Oscillator power supplies can be considered to be essentially at ground potential, for batteries or A.C. power packs have a fairly high capacity with respect to ground. If the oscillator is not "tied down" (connected) to the same ground as the power supply, anything coming near the oscillator chassis and any slight circuit change may cause undesirable changes in frequency and power output. The power supply circuits for the various tube electrodes may introduce undesirable paths for the A.C. currents; filters and chokes are therefore used extensively to keep A.C. power where it belongs. Each type of oscillator can be fed with power from the power supply either through some signal component (known as the series feed method) or directly through a choke which keeps signal currents out of the supply (known as the shunt or parallel feed method). To get definite phase relationships the plate, the cathode or the grid may be tied to ground.

Oscillators shown in Figs. 6A, 6B, 6C, 6D, 6G, 6H, 6I, 6J, 6K, 8A, 8Cand 8D have series fed plate supplies; note that in each case the D.C. plate current flows through a circuit part which is essential for oscillation. Oscillators shown in Figs. 6E, 6F and 8B are parallel or shunt fed, the D.C. plate current being fed through an R.F. choke for high frequencies, and through an A.F. choke for low frequencies. The choke allows the plate to assume an A.C. potential without feeding A.C. into the plate supply.

A study of the power supply connections in the various forms of the widely used Hartley oscillator will bring out some important principles which apply to all oscillators. Although the circuits shown in Figs. 8A, 8B, 8C and 8D may at first glance appear different, they are all Hartley oscillators because the tank coil is in both grid and plate circuits in each case, with the mid-tap of the tank coil connected to the cathode either directly or through a condenser to divide the two circuits.

Plate Grounded. The Hartley circuit in Fig. 8A contains a filament type tube with the plate connected directly to ground; the grid and cathode

alternately acquire positive and negative potentials, which are always out of phase with each other, and oscillation is therefore maintained. Since the filament must receive D.C. power from its supply, a choke is introduced to make the filament assume an A.C. potential and to keep the A.C. signal from feeding into the supply. The use of a heater type tube would eliminate the need for the cathode choke, as then the cathode would definitely be isolated from the A supply.

Filament Grounded. Figure 8B shows another filament type tube in which the filament is fed with A.C. and has its mid-point tied to ground. Feed-back condenser C provides an A.C. path between grid and plate circuits but blocks the flow of D.C. current to ground. The tank coil center tap is grounded directly to the mid-tap of the filament resistor; each section of this resistor is shunted by a condenser to allow signal currents to flow to the filament proper.



FIG. 8. Four methods of making ground connections in Hartley oscillator circuits are shown here. The frequency-controlling circuit is shown in heavy lines in each case.

Grid Grounded. Rarely is the grid grounded, but if it were necessary, the circuit shown in Fig. 8C could be used. As far as A.C. signals are concerned, the grid is tied to ground through C_1 (shunting the grid resistor), and the mid-tap of the tank coil is connected to the cathode through condenser C_2 , which prevents D.C. from flowing back to ground. The cathode circuit choke coil *RFC* allows the cathode to assume an A.C. potential without interfering with the flow of direct current from ground to cathode to plate to B+.

Floating Ground. Figure 8D illustrates a rather well-known circuit, often called the *push-pull or push-push Hartley oscillator*, in which a floating R.F. ground connection to the tank coil is used. Tap X is placed as close as possible to the electrical center of its coil, making the voltages

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across each half of the coil practically equal at all times and opposite in polarity. But these voltages must be *exactly* equal, each section of the double triode tube getting equal excitation and equal plate currents, if the optimum operating condition is to be obtained; here is where choke RFC enters into the picture. Any slight difference in these tank coil voltages results in a small-R.F. voltage drop across the choke, and this drop has the effect of swinging point X to the exact electrical mid-point of the tank coil.

FREQUENCY STABILITY

When an oscillator circuit is adjusted to a desired frequency shortly after it has been placed in operation, and a check of frequency is made an hour or so later, you will generally find that the frequency has changed or *drifted*. This is highly undesirable, for an oscillator should maintain as near constant frequency as possible. There are three causes for frequency drift in a self-excited oscillator: 1, Changes in the electrical values of parts in the oscillator circuit; 2, changes in load; 3, changes in tube constants, due to power supply voltage changes or to the effects of causes 1 and 2.



FIG. 9A. Coil *L* has been added to this Colpitts oscillator to minimize frequency drift which is caused by variations in the spacings between electrose as tube temperature varies.

FIG. 9B. The insertion of resistor R_1 in this tuned plate oscillator improves the frequency stability of the circuit, although its use cuts down the output power.

Assuming that sturdy, well designed coils, condensers and resistors which will not change in value of their own account are used, the greatest amount of frequency drift will be caused by *electrical changes* due to temperature changes; room temperature may fluctuate, and parts may heat up because of current passing through them. The solution lies in selecting parts on which temperature has little or no effect; if this proves insufficient, those parts which are most sensitive to temperature can be placed in a temperature-controlled oven.

Effects of variations in load can be reduced by keeping the coupling to the load as loose as possible or by using an ordinary amplifier, often called a *buffer* in this particular case, to separate the oscillator tube from its load.

Changes in tube constants are not so easily remedied; the A.C. grid-tocathode resistance (R_z) and the A.C. plate-to-cathode resistance (R_p) have been mathematically proven to be the important factors. For a given oscillator tube and circuit the value of $R_z \div R_p$ is always a definite value regardless of the condition in the circuit; keeping the change in one as low as possible thus keeps the other one fixed and insures good stability. Frequency drift is further reduced by using a low loss tank circuit (with high Q), which in turn calls for a small load and a low loss coil and condenser. The use of high values of grid bias resistors (but not so high that they block the circuit) makes $R_{\rm g}$ assume a high and practically constant value. If a series resistance of high value is connected between the plate and the B supply (or between the plate tank circuit, if used, and the B supply), this resistor will tend to absorb any changes in plate voltage, leaving $R_{\rm p}$ constant.

When high Q tank circuits are used, the frequency can be made practically independent of the values of R_p and R_g by inserting coils or condensers (or both) in series with the grid or plate leads (or both) of the tube. An example is shown in Fig. 9A for a Colpitts oscillator; the electrical value required for the added coil L is determined by the values of C_1 , C_2 and L_1 .



Top row: Three views of a General Radio type 376-L quartz plate (crystal) holder. The quartz plate itself can be seen inside its fiber mounting ring in the view at the left. This unit is of the plug-in type: one prong connects to a metal electrode beneath the quartz plate, while the other prong makes contact with the machined metal cover (top center) which serves as the upper electrode. Lower left: Natural crystals of Brazilian quertz, from which quertz plates used in crystal oscillators will be cut. (Photo supplied through courtesy of Bliley Electric Co.). Lower right: A Western Electric quartz plate in its holder; this unit is also of the plugin type. The quartz plate is here held in position between the two metal electrodes by a spring-steel plate bent at a right angle.

A more practical method, intended for use in cases where the tube capacities are relatively unimportant, is given in Fig. 9B; there is a low loss (high Q) tank circuit (LC), a high resistance $(R_1)^*$ in the plate-totank circuit to minimize the effects of voltage and A.C. plate resistance fluctuations, and a small load. R_2 being much higher in ohmic value than R_3 and both being high with respect to the A.C. plate resistance, changes in the load have very little effect on circuit stability. This circuit sacrifices power for good frequency stability. In general, however, any oscillator

* R, should not be so large that it stops oscillation.

should be allowed to heat up from one to two hours before use, where minimum frequency shift is essential.

CRYSTAL OSCILLATORS

Frequency Stability Requirements. In radio transmission, frequency stability becomes of extreme importance. Each year-governmental agencies, such as the Federal Communications Commission in the U. S. A., are demanding that frequency drift be reduced and that the exact frequency be nearer to the assigned station value. The present permissible drift is 20 cycles in the broadcast band; if a certain station operates at 1,000,000 cycles, this means the transmitter carrier frequency drift must be less than .005 per cent. Frequency-monitoring equipment used in the station must have even greater stability than this. It is difficult to design self-excited oscillators which will hold their frequency this closely but fortunately the quartz crystal, when properly cut and ground to size, mounted in a holder located in a constant temperature oven and connected into a tube circuit, gives an oscillator which has acceptable frequency stability.

Types of Crystals. Crystals are solids and therefore have height, width and depth; lines parallel to these three dimensions of a crystal are called the crystal axes. There is the X or electrical axis, the Y or mechanical axis and the Z or optical axis, all at right angles to each other. By cutting a quartz crystal into small slabs across the X axis, a so-called X-cut crystal is obtained; by cutting across the Y axis a Y-cut crystal is obtained; Z-cut crystals are of little importance in radio. When one of these crystals is placed between two metal plates, one surface of the crystal being about .003 inch away from one of the plates to allow the crystal to vibrate freely, an A.C. voltage applied to the two plates will cause the crystal to vibrate at a frequency determined chiefly by the thickness of the cut crystal. For an X-cut crystal the frequency in cycles is approximately 3,000,000 divided by the crystal thickness in millimeters; for a Y-cut crystal the frequency in cycles is about 2,000,000 divided by the thickness in millimeters. Crystals which vibrate up to 5,000 kc. are quite easily made.

The really important feature of a crystal, as used in a crystal oscillator circuit, is that it acts as a tank circuit which has a better frequency stability than any other practical device; only changes in crystal temperature produce frequency drift, and temperature can be easily controlled.*

Study of a Crystal Oscillator Circuit. The simple crystal oscillator circuit given in Fig. 10A is essentially the circuit of Fig. 6H with a crystal in place of the grid tank circuit. This crystal is cut for a certain desired frequency, and will oscillate when the plate tank circuit is made inductive (by tuning L-C to a frequency slightly higher than the resonant frequency of the crystal). The plate tank circuit then has the effect of feeding back to the grid circuit, through the grid-to-plate capacity of the tube, a voltage which is in phase with the crystal voltage and which therefore serves to

^{*}When quartz is cut at an angle to both the Y and Z axes, giving what is known as a V or A.T. cut crystal, the resulting crystal is little affected by temperature changes. V-cut crystals are used where temperature-controlled ovens are not justified because of space or weight limitations.

maintain the vibrations of the crystal. Figure 10B shows that as condenser C is varied from a low to a high value, oscillation starts at a certain value of C (point 1) and plate current decreases to a minimum at point 3 as the value of C is further increased. As C is increased beyond point 3, plate current suddenly rises to its original value, indicating that oscillation has stopped.

Operating Point. Although best efficiency is obtained when D.C. plate current is a minimum (at point 3), the circuit is not stable at this operating point (slight increases in C will stop oscillation); point 2 is the most practical operating point, for here changes in load or plate voltage have less effect.



F1G. 10A. Simple crystal oscillator circuit. Rectified grid current flowing through grid resistor R_0 provides the correct bias for the tube, while condenser C controls the starting, stopping and intensity of oscillation.





Crystal oscillator circuits are of many forms, the greatest variation being in the coupling to the load; electron coupling is widely used. Crystal oscillators have a very low output power, and R.F. amplifiers must therefore be used in practically every instance to get sufficient power output even for test equipment.

ULTRA HIGH FREQUENCY OSCILLATORS

Crystal Oscillators are Unsatisfactory for U.H.F. Work. Oscillators which must generate very high radio frequencies with very low frequency drift introduce many special problems. To be sure, crystal oscillators can be used with one or more special harmonic-producing R.F. amplifiers which are adjusted to double and redouble the oscillator frequency until the desired ultra high frequency (u.h.f.) is obtained. A single amplifier operated with a C bias greater than the plate current cut-off value will, if excited with a signal of some definite frequency, produce in its plate circuit the 2nd, 4th, 6th, 8th, etc. harmonics of this input frequency, the 2nd harmonic being the strongest. By using other similar amplifiers, each tuned to the 2nd harmonic of the preceding stage, the original frequency produced by a crystal oscillator can be doubled many times, but there are serious drawbacks to this doubling method. When the crystal frequency is 5,000 kc. (the practical maximum operating frequency of a quartz crystal), and frequencies of the order of 100,000 kc. are desired (such as for television purposes), it is expensive as well as difficult to build frequency doublers which will operate satisfactorily at such high frequencies.

Self-Excited U.H.F. Oscillators. Acorn type tubes are used extensively with very high Q tank circuits for low power, self-excited u.h.f. oscillators. The tank coils are wound with stiff, solid wire which requires no supporting form, thus keeping the distributed capacity and the losses of the coil at a minimum. Midget tuning condensers can be used, but often the required tank capacity is obtained by varying the spacing of the coil turns. This form of u.h.f. oscillator is used in television receivers.

Tank Coil Voltage Distribution. When the mid-tap of the tank coil in a self-excited u.h.f. oscillator is grounded as in Fig. 11A, the coil ends are alternately positive and negative^{*}; when one end of the coil is grounded, as in Fig. 11B, the other end likewise changes in polarity. The distribution of voltage is sinusoidal, hence Fig. 11A shows half-wave $(\lambda/2)$ and Fig. 11B shows quarter wave $(\lambda/4)$ voltage distribution. This does not necessarily mean that the coil itself is a half or quarter wave length long in physical size; the physical and electrical lengths of the coil can be made the same, however, by using straight metal wires or pipes, as shown in



2FIG. 11. The curves shown here give the voltage distribution across: A, the coil of an oscillator tank circuit when the coil mid-tap is grounded through an R.F. by-pass condenser; B, the coil of an oscillator tank circuit when one end of the coil is grounded; C, one of the pipes in a u.h.f. tank circuit using two parallel or concentric metal pipes. In each case the horizontal reference line represents ground

potential or zero voltage; in each case the curve surrounding a shaded area represents conditions for that half of a cycle when the lefthand end of the tank coil is positive, and the other curve (the minus curve) portrays conditions for the other half of the cycle. The curves at C are for one pipe (No. 1) only; curves for the other pipe would have exactly the same shape but opposite polarity.

Fig. 11C. Here pipes 1 and 2 are separated from one to four inches by air. When very high frequency, low wave length circuits are desired, simple pipes are used without tuning condensers.

High Power U.H.F. Oscillators. For high power u.h.f. oscillators, coile are made of copper tubing; sometimes two straight copper pipes side by side or one inside the other are used to provide the required inductance and capacitance. A single turn of tubing or even less is generally sufficient. Push-push tube circuits like that shown in Fig. 12A are customarily used; to get close grid-to-plate coupling (unity coupling), the grid wire is run inside the length of copper tubing which connects the plates together. Notice how electrode voltages are fed to the mid-points of the plate and grid loops or coils.

The circuit shown in Fig. 12B is a tuned grid, tuned plate oscillator (essentially the same as the fundamental Armstrong circuit shown in Fig. 6H), but using parallel pipes which may or may not be concentric. So-called quarter wave length lines are used to tune the grid and plate circuits. Since the grid tank circuit governs the frequency, it should be mechanically designed so that temperature changes do not affect its physical length. The

^{*}The coil tap can be grounded directly or through a condenser; either procedure places the coil tap at zero A.C. potential with respect to ground.

plate tuned line can be replaced with a coil and condenser (or coil alone) to conserve space, for it merely needs to be tuned slightly inductive.

Fig. 12C is the ultra-audion circuit shown in Fig. 6F with parallel lines substituted for a coil-condenser tank circuit; whereas the grid and plate in Fig. 12B were tapped along the line and the filament was connected to the end of the line, in Fig. 12C the grid and plate are tapped opposite each other along the line. The latter method gives twice the A.C. tank voltage. The taps are made variable so a better impedance match can be obtained, giving better circuit efficiency and stability. Points 1 and 2 in either circuit are load-coupling points and are usually variable as to position.

The three circuits in Fig. 12 are widely used in television transmitters, where high power and good frequency stability are required. They will operate at from 50 to 200 megacycles, the higher frequencies generally



FIG. 12. Three high power ultra-high frequency oscillator circuits which are widely used in television transmitters.

being used to relay a program from a pick-up point to the main studio or from studio to transmitter. Amateurs use these circuits for communication in the 5 meter (56 megacycle) band. Air core coils and midget variable condensers are preferred in television receivers, although lines are occasionally used where space permits.

DYNATRON OSCILLATORS

When the voltage applied to a resistor is increased, we naturally expect the current through the resistor to increase. Likewise, when the plate voltage of a tube is increased (the grid bias being at less than cut-off), we expect that plate current will increase. But under certain circumstances, especially with vacuum tubes, it is found that the current actually *decreases* when the voltage is increased. This can mean only one thing, that somehow the vacuum tube is feeding energy back into the circuit. A circuit acting like this is said to have *negative resistance*.* Naturally, when an

^{*}By definition, resistance is voltage divided by current. A generator can be looked upon as a resistor whose ohmic value is equal to the generated voltage divided by the current flowing; since this voltage is equal and opposite to a real or positive voltage drop in a resistance, the generated voltage has a *negative resistance* effect; it cancels the effects of real resistance, supplying the power required by the real resistance.

oscillator tank circuit is connected to terminals which are acting as generator terminals and which have this negative resistance effect, the oscillator circuit will be fed with energy to overcome its resistance loss and oscillation will take place.

A Dynatron Oscillator Circuit. A screen grid tube is one device which can act as a negative resistance. When the plate is at a lower positive voltage than the screen grid, considerable secondary emission will take place at the plate, and these secondary electrons will be attracted to the screen. An increase in the plate supply voltage (but not above the screen grid voltage) increases the secondary emission, thus making the plate current decrease; the screen therefore gets more electrons than the plate. A typical oscillator circuit using a screen grid tube is shown in Fig. 13A; this is known as a dynatron oscillator circuit. For a given C bias voltage (determined by the setting of P_2) a plate voltage-plate current characteristic like that in Fig. 13B is obtained. Notice that after E_P reaches about



FIG. 13. A dynatron oscillator circuit (A) and its plate-voltage plate-current characteristic (B).

17 volts, further increases up to about 34 volts actually reduce $I_{\rm P}$. This means that when the plate voltage of a screen grid tube varies within the shaded region in Fig. 13B, the tube will behave as if it had a negative resistance equal in ohmic value to $v_{\rm n}$ divided by $i_{\rm n}$.

In the dynatron circuit of Fig. 13A this negative resistance characteristic makes the tube act as a generator which supplies energy to the L-C tank circuit to overcome resistance losses. Oscillation will occur in a dynatron circuit whenever the resonant resistance of the L-C tank circuit is numerically greater than the negative resistance of the tube. The minimum oscillator output voltage, v_n , is obtained when the two resistances are practically equal; increasing the resonant resistance makes the oscillator output voltage swing beyond the shaded area in Fig. 13B. Oscillation is most stable when the D.C. plate voltage is in the middle of the region of negative resistance. A dynatron oscillator delivers a nearly pure sine wave voltage to a load which is inductively coupled to the tank circuit coil.

Oftentimes we get dynatron oscillator action in a vacuum tube circuit where it is not desired, resulting in squeals. The undesired oscillations are then known as *parasitic oscillations*; they can be stopped by increasing the D.C. plate voltage to get it beyond the region of negative resistance in Fig. 13B, by adjusting the C bias to get this same result, or by inserting in the plate lead of the tube a resistor which will make the negative resistance of the tube too high for oscillation to occur.

RELAXATION OSCILLATORS

The cathode ray tubes used in television pick-up cameras, in television picture reconstructors and in cathode ray oscillographs used for testing purposes, require a special kind of oscillator, one which will produce a "saw tooth" voltage or current characteristic like that shown in Fig. 14A. This saw tooth wave form is needed to sweep the electron beam at a uniform rate across the cathode ray tube screen, then return it almost instantly to the starting point for another sweep. Oscillators capable of producing a saw tooth wave form are known as sweep oscillators or as saw tooth oscillators. In radio circuit testing work, oscillators capable of producing the square top wave form shown in Fig. 14B are required, for this wave form results in strong harmonics.

In general, circuits for these saw tooth and square top oscillators use a condenser which is alternately charged up by a D.C. voltage and discharged by an electrical device such as a gaseous glow tube. The output voltage drops to zero or *relaxes* for a definite fraction of each cycle, and such units are therefore called *relaxation oscillators*.





TIME TIG. 14B. One type of relaxation oscillator FIG. 14B. produces this sam-tooth wave form, which lator prod gives the sweep voltege required for laboratory cathode ray tubes.



FIG. 14B. Another form of relaxation oscillator produces this square-top wave, used in laboratory oscillators because it produces very strong harmonics.

The simplest circuit for generating a saw tooth voltage is given in Fig. 15A. In this circuit N is a neon glow tube consisting of two electrodes in a glass envelope filled with neon gas. This tube will not conduct electricity until a definite voltage, called its *firing* or *ignition* voltage, is applied; the tube resistance then becomes very low and a large current can flow unless it is limited by a resistance. Thus, when a condenser C is charged through resistor R, the condenser voltage gradually builds up (depending on the time constant; R in megohms times C in microfarads gives time in seconds), until the voltage across C is high enough to ignite the neon tube. The resulting flow of current through N lowers its resistance materially, thus shorting the condenser and dropping its voltage practically to zero (most of the line voltage being taken by R); the neon tube then stops glowing or firing and the action starts over again. The frequency is controlled by the characteristics of N, the values of C and R, and the value of the D.C. supply voltage.

Gaseous Triode Circuit. A gaseous triode tube is often used with a condenser-resistor charging circuit to produce a saw tooth wave, it being easier to control the firing voltage of this tube than of a diode. In Fig. 15B, T is a Thyratron tube (mercury vapor-filled) or a grid glow tube (filled with neon gas). Conduction through the tube depends upon the ratio of plate to grid voltages; the more negative the C bias, the higher the plate voltage must be before the tube will conduct current. Once conduction starts, only the reduction of plate voltage to a very low value will stop current flow. In this circuit the condenser voltage builds up gradually through R until sufficiently high to fire the tube; the condenser is momentarily short-circuited by the tube as it discharges, the tube stops conducting and the charging of C starts over again. The frequency is essentially controlled by the values of C and R, for the plate and grid voltages are usually adjusted to correspond to the desired time constant. A controlling signal, such as the line or picture frequency of a television-broadcast, can be introduced at point X, if the C bias is made sufficiently high so that the tube will discharge only when the controlling impulse is applied. The circuit can thus be made to produce a saw tooth signal output which is in synchronism with an input signal.



FIG. 15. Relaxation oscillator circuits.

An electron-coupled vacuum tube amplifier may be used instead of the gaseous tube signal circuit; a screen grid tube circuit is shown in Fig. 15C. Space (cathode) current is made to flow by applying a positive voltage to the screen and a negative C bias voltage to the grid. Only a negligibly small current flows to the plate until it is placed at a very high voltage; at that point the plate-to-cathode resistance drops to a low value. Again a condenser charged through a resistor is used to build up a voltage which will break down the resistance of the tube (which is in parallel with the condenser).

In the circuits of Fig. 15A, 15B and 15C, the desired saw tooth voltage is obtained by connecting the load across condenser C. A resistor of high ohmic value (10 to 30 times the value of R) is placed in series with the load to insure a uniform build-up of voltage across C. Quite often, charging resistor R is replaced by a vacuum tube, the cathode-to-plate resistance of the tube serving as a self-regulating high resistance. The tube has the characteristic of making the condenser charge more evenly with time; in either case the presence of the charging resistor prevents the condenser from being charged to line voltage.

The Multivibrator. This important type of relaxation oscillator, which produces the square top wave form shown in Fig. 14B, depends for its

operation upon the fact that the input and output voltages of a resistancecoupled amplifier are 180 degrees out of phase. The multivibrator circuit consists of two amplifiers connected as in Fig. 15D, each amplifier using one section of the double triode tube.

The operation of this circuit is a sort of "see-saw" affair in which first one tube and then the other passes plate current to produce the desired square top wave. When the circuit is first placed in operation, an increase in plate current in one tube causes a decrease in plate current in the other (because the two stages are out of phase with each other), and this initial increase is reamplified almost instantly to make the current in that tube rise to a maximum while the plate current in the other tube drops to



zero. Current continues to flow through the tube, forming one square-top portion of the wave, until such time as leakage through the resistor and condenser cause a slight plate current increase in the non-conducting tube; this increase is almost instantly amplified and reamplified, forcing current in the first tube down to zero. The time interval for which the current flows in each tube depends upon the values of R_1 , R_2 , R_3 , and C, unless an A.C. control voltage is applied to one of the grids to set the rate of oscillation. This controlling signal may be the fundamental frequency or any harmonic of the natural frequency of the multivibrator; the input signal, which may come from a crystal oscillator, therefore controls the stability of the multivibrator.

AUDIO AND BEAT FREQUENCY OSCILLATORS

Oscillators which produce signals in the aural band (from about 30 to 16,000 cycles) differ in no way from R.F. oscillators; it is merely a matter of getting a sufficiently high inductance and capacity in the tank circuit. For a 100 cycle oscillator the inductance should be about 25 henrys if the capacity is .1 mfd., and 2.5 henrys if the capacity is 1 mfd. Air core coils

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with inductances of this order would be enormous. Saturation in iron core coils introduces distortion, but if an appreciable air gap exists in the magnetic path an A.F. signal which is entirely satisfactory for radio testing (defect isolation) purposes can be had.

A typical Hartley audio frequency oscillator circuit is shown in Fig. 16A. The iron core coil can be the output transformer of a push-pull amplifier. Grid condenser C_1 must be rather high in value, for it must offer a low reactance to low frequency audio signals. Tuning is accomplished by varying C_3 , the maximum frequency being governed by the distributed capacity of the transformer windings.

Beat Frequency A.F. Oscillators. Iron core coils having a low distributed capacity and operating below saturation are costly and even difficult to build in units which will cover satisfactorily the entire A.F. band with a single tuning condenser. For this reason audio oscillators generally employ the beat between two low R.F. signals. For example, if a 100 kc. and



FIG. 17. Schematic circuit disgram of a typical modulated R.F. oscillator,

a 102 kc. signal are fed to a detector, the 2,000 cycle difference frequency will appear in the plate circuit. If the load is coupled to the detector plate circuit with an iron core transformer having a condenser shunted across its primary, the two fundamental frequencies and the 202 kc. sum frequency will automatically be shorted out or by-passed. Beat A.F. oscillators are widely used for testing and for precision checking work. The great difficulty lies in preventing the two R.F. oscillators from drifting in frequency.

The circuit of a beat frequency oscillator, built from parts found in the average radio shop, is given in Fig. 16B. Each R.F. oscillator should be placed in an individual shielded compartment. With C_{AF} set at minimum capacity, adjust either of condensers C until the D.C. plate current in the detector is at its no-excitation value; this places both oscillators at the same frequency and there will therefore be no audio output.

Calibration of this oscillator is rather difficult with ordinary equipment, but a rough calibration can be obtained by comparing the audio beat output notes for various settings of C_{AF} with notes produced by tuning forks of known frequency. For precision results, it is far better to buy a commercial A.F. oscillator than to build your own.

A SERVICEMAN'S MODULATED TEST OSCILLATOR

All of the oscillators discussed so far produce constant amplitude A.C. signals. In testing radio receivers, however, the modulated radio signal ordinarily picked up must be replaced by an equivalent test signal. An R.F. oscillator which is modulated with an audio or video signal is, therefore, needed by a serviceman. There are many ways of producing this, but in a test oscillator (or signal generator) the easiest way to get this is to vary the D.C. plate voltage applied to the R.F. oscillator by introducing an A.C. voltage of an audio or video frequency. Figure 17 shows a typical modulated R.F. oscillator extensively used by servicemen; the low frequency or modulating signal is produced in the $V_{\rm LF}$ oscillator, the output from transformer T being connected in series with the D.C. plate supply to tube $V_{\rm RF}$ in the R.F. oscillator circuit. The degree or percentage of modulation is varied by varying R_3 , which changes the amount of low frequency voltage.

The low frequency oscillator could just as well be replaced by an electric phonograph reproducer whose output was fed to points 1 and 2. For testing all-wave receivers, coils L-L should be of the plug-in type; otherwise an arrangement whereby a switching mechanism inserts the proper coils must be used. Calibration of the oscillator is quite a tedious and exacting process, and since test oscillators are inexpensive, commercial equipment is always preferred.

TEST QUESTIONS

Be sure to number your Answer Sheet 21FR-2.

Place your Student Number on every Answer Sheet.

Never hold up one set of lesson answers until you have another ready to send in. Send each lesson in by itself before you start on the next lesson. In this way we will be able to work together much more closely, you'll get more out of your Course, and you will receive the best possible lesson service.

- 1. In a vacuum tube oscillator, are the power losses in the L-C oscillatory circuit compensated for constantly or intermittently?
- 2. What, in a self-excited vacuum tube oscillator, essentially governs the frequency?
- 3. Give another name for the oscillatory circuit.
- 4. In the oscillator circuit of Fig. 4, what two parts automatically supply a steady D.C. bias voltage to the grid?
- 5. What happens if the ohmic value of the grid resistor in a self-excited vacuum tube oscillator (such as R_c in Fig. 4) is made too high?
- 6. When a load is applied to the tank circuit of a self-excited vacuum tube oscillator, will the D.C. plate current *increase* or *decrease*?
- 7. Can the power output of a self-excited vacuum tube oscillator be increased by increasing the D.C. plate supply voltage?
- 8. How can the Hartley oscillator circuit be identified?
- 9. What is the purpose of the parallel resonant plate circuit in the tuned grid, tuned plate Armstrong oscillator?
- 10. To what part of a self-excited oscillator circuit should the load be coupled in order to feed it with the fundamental oscillator signal?

World Radio History

THOROUGHNESS

Whatever you do, do well if you would stay on the straight road to success. The habits of carelessness and slipshod work are all too easy to acquire; beware of them as you would the plague. Men who are thorough in their work cannot remain undiscovered for long, because the demand for such men is greater than the supply.

Thoroughness is just as important in study as it is in work; what you get out of a lesson depends upon how completely you master the material presented in it. Some books, as fiction, are read hurriedly and only once, then cast away; the enduring works of literature are carefully read and reread many times but always essentially for the pleasure they give; textbooks, however, must be read quickly, to get the basic ideas, then carefully many times until every important principle has been mastered. Textbooks are always saved for future reference; tomorrow you may have urgent need for the information given in a paragraph which today seems so insignificant.

Thoroughness in study habits leads to thoroughness in work habits, and eventually to a thorough success.

J. E. SMITH

PEAK AND BAND-PASS R. F. TUNING CIRCUITS

David

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NATIONAL RADIO INSTITUTE ESTABLISHED 1914 WASHINGTON, D. C.



STUDY SCHEDULE No. 22

For each study step, read the assigned pages first at your usual speed. Reread slowly one or more times. Finish with one quick reading to fix the important facts firmly in your mind then answer the Langer Operations for these series
Study each other step in this same way.
I. Importance of Response Curves; General Analysis of a Modulated R.F. Carrier; Analyzing Typical Response Curves
This general information on response curves prepares you for speedier under- standing of the actual tuning circuits taken up in the rest of the lesson. Study this section thoroughly. Answer Lesson Questions 1, 2 and 3.
2. Factors Controlling Response; Coupling Methods for R.F. Tuning Circuits Pages 6-11
Here is where you hegin studying in detail exactly how coils and condensers work together to permit passage of the desired signal while rejecting all other signal frequencies. Answer Lesson Questions 4, 5 and 6.
3. Directly Coupled Resonant Loads; Tuned Secondary Transformer LoadsPages 11-16
This section should not prove at all difficult if you realize right from the start that your goal now is simply understanding the material. Don't try to memorize it, because you can always review this material later, and can refer to it when in need of specific facts. Answer Lesson Question 7.
4. Double-Tuned Transformer Loads; Double-Tuned Capacity-Coupled Loads
5. Combination Capacitive and Inductive Couplings; Tuned Secondary with Fixed Tuned Primary Circuit; R.F. Tun- ing Circuits Used in Actual Receivers Pages 23-28 Pay particular attention to the section covering a fixed tuned primary circuit, because this scheme is used in a great many table model receivers. The actual tuning circuits at the end of this lesson illustrate many of the important principles previously studied.
6. Mail Your Answers for This Lesson to N.R.I. for Grading.
7. Start Studying the Next Lesson.
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1948 Edition

Peak and Band-Pass R. F. Tuning Circuits

Importance of Response Curves

PRACTICAL radio men are today more concerned than ever before with the shapes of the resonant response curves for R.F. amplifiers, for they have come to realize that these curves reveal the exact characteristics of a receiver or transmitter and tell when undesirable effects have been insired gain and fidelity within the limitations of the tuning circuits. A thorough understanding of the peculiar characteristics of R.F. tuning circuits and an ability to read the story told by each shape of response curve will prove particularly valuable when using a cathode ray oscilloscope for radio receiver testing and servicing.

You are already familiar with peak



All-wave superheterodyne receiver being aligned for band-pass response, using a frequencywobbulated R.F. signal generator (extreme left) and a cathode ray oscilloscope. The final double-peak response curve, secured after all adjustments are made, can be seen on the screen of the cathode ray tube. Note that the receiver chassis is set on end, for convenience in making connections and adjusting under-the-chassis trimmer condensers.

troduced by adjustments or by defects in circuit parts.

Older receivers, as well as a great many modern receivers, use R.F. tuning circuits which are adjusted for peak response. On these receivers a serviceman need only adjust for maximum output, never giving a thought to the shape of the peak response curve. A modern high-fidelity radio receiver has band-pass R.F. tuning circuits, however, and actual viewing of the response curve greatly simplifies the adjusting of the receiver to give the deresponse curves like those shown in Fig. 1A, for they have been discussed in previous lessons. You know that a sharp peak response curve for an R.F. amplifier indicates high gain and high selectivity, while a broad peak response represents somewhat lower gain and lower selectivity but better fidelity. Likewise you are familiar with the band-pass response curves shown in Fig. 1B, and know that R.F. tuning circuits having these curves give better fidelity at the expense of gain. In this lesson we will study in detail the tuning circuit conditions which give to an R.F. amplifier any of these four response curves or any of the many possible variations of these curves.

General Analysis of a Modulated R.F. Carrier

If we used a special cathode ray oscilloscope to analyze an R.F. carrier which is 100% modulated with a single sine wave signal (of frequency f_m), we would see on the screen of the cathode ray tube a pattern much like that in Fig. 2A (the dotted lines indicating the modulation envelope would of course be absent). Either a mathematical analysis or actual measurements will



FIG. 1. Typical peak and hand-pass response curves of R.F. amplifiers.

show that we really have three different R.F. signal frequencies in this modulated carrier, as indicated in Fig. 2B:

f, the R.F. carrier frequency

- f1, the lower side frequency, which is equal to the carrier frequency minus the modulation frequency $(f_1 = f_m)$
- f_2 , the upper side frequency, which is equal to the carrier frequency plus the modulation frequency $(f_2=f+f_m)$

Furthermore, with 100% modulation the voltage of each side frequency signal will be *exactly one-half* that of the carrier signal. (With less than 100% modulation, the amplitude of each side frequency will be *less than one-half* that of the carrier.) In dealing with R.F. tuning circuits, we must consider all three of these R.F. signals, for the side frequencies must be amplified the same amount as the carrier frequencies if distortion is to be avoided.

When a 100%-modulated R.F. carrier is sent through an R.F. amplifier which has a perfectly flat top response, the side frequencies will be amplified equally as much as the carrier and the output wave pattern will be identical to the input wave pattern. If, however, we send this 100%-modulated R.F. carrier through a tuned R.F. amplifier which is considerably off tune, severe amplitude distortion occurs because the side frequencies and the carrier are amplified different amounts, and the output wave pattern might be as shown in Fig. 2C (this pattern corresponds to the condition where one side frequency is not amplified at all and the other side frequency is amplified twice as much as the carrier; Fig. 2D indicates the output voltage relationship under this condition). Output wave patterns thus tell directly whether distortion is occurring in an R.F. amplifier.

Unfortunately the average cathode ray oscilloscope used by servicemen is not designed to amplify R.F. carrier voltages sufficiently to give useful modulated R.F. patterns on the screen; an extra R.F. amplifier would have to be used, or a costly and bulky laboratory type oscilloscope secured. A response curve of the R.F. amplifier in question gives essentially the same information about distortion, however, and is easily produced with an ordinary radio servicing oscilloscope. A typical peak response curve is shown in Fig. 2E; this curve tells how much amplification the side frequencies will get at any modulation frequency value. Each response curve has a story to tell you; to show how these stories can be read, we will consider a typical example in which the carrier frequency is assumed to be 1,000 kc., with 100% modulation.

When the modulation frequency is 100 cycles, the side frequencies will be 999.9 kc. and 1,000.1 kc.; by referring to Fig. 2E, where these side frequencies are designated as f_3 and f_4 , we can readily see that these will receive essentially the same amplification (gain) as the 1,000 kc. carrier. This means that after the modulated signal has passed through the R.F. amplifier, the two side frequencies will each be the same fraction of the carrier voltage (one-half in this case of 100% modulation) as they originally were.

With a 5.000-cycle modulation signal, however, the resulting 995 kc, and 1.005 kc. side frequencies (f_1 and f_2 in Fig. 2E) receive considerably less amplification than the carrier (f): this means that after the modulated signal has passed through the R.F. amplifier. the two side frequencies will be a considerably lower fraction of the carrier voltage than they originally were (each will be less than one-half the carrier voltage in our case of 100% modulation). With only one modulation frequency, this attenuation of side frequencies is simply equivalent to a reduction in the modulation percentage, provided that both side frequencies are equally attenuated; after dethen, the 5,000-cvcle modulation. audio signal voltage will be lower than if there were no attenuation of side frequencies.

Frequency Distortion. When a number of different modulation frequencies ranging from 0 to 5,000 cycles (such as we have in radio receivers which are tuned to sound broadcasts) are present in an R.F. amplifier having the response curve in Fig. 2E, there will be a large number of side frequencies in the range from 995 kc. to 1,005 kc. Those farthest away from the carrier frequency, corresponding to the higher modulation frequencies, will be amplified the least, with the result that a certain amount of frequency distortion will be present in the audio signal after demodulation. If not too severe, this frequency distortion can be corrected in a receiver by the use of equalizing circuits which make the audio amplifier provide increased amplification for those higher modulation frequencies which were cut down in the R.F. tuning circuits. Servicemen often use this little equalizing trick to com-



FIG. 2. These diagrams tell you what happens to a modulated R.F. carrier signal when the R.F. amplifier is properly tuned (A, B and E) and when it is improperly tuned (C, D and F.)

pensate for frequency distortion in a highly selective R.F. amplifier. In television circuits the modulation frequencies may range from 0 to over 2.5 megacycles, and the amount of equalization required in picture signal amplifiers may therefore be quite great.

Amplitude Distortion. When the side frequencies associated with an R.F. carrier are attenuated or cut down unequally by a tuning circuit (so that with a single modulation frequency, one side frequency will have a greater amplitude than the other), *amplitude distortion* as well as frequency distortion will be present. This fact is not so easily shown without mathematics, but by considering the extreme case where only one side frequency is allowed to pass, the other being cut out entirely by the tuned circuit, we can get some idea as to why this statement holds true.

Suppose that the R.F. amplifier in our previous example is tuned to 1,000 kc. but is fed with a 995 kc. carrier modulated at 5,000 cycles (5 kc.); now we have the condition represented by Fig. $\mathcal{Z}F$, where the upper side frequency (1,000 kc.) is fully amplified, the carrier is amplified about half as much, and the lower side frequency receives hardly any amplification at all. The R.F. amplifier under this condition allows only one side frequency to pass through with the carrier, as was indicated in Fig. 2D; both will have the same amplitude at the output, and the wave form of the amplifier output voltage will be as shown in Fig. 2C. (Originally, as in Fig. \mathcal{B} , the amplitude of the carrier f was twice that of the upper side frequency f_2 ; reducing the carrier amplitude one-half without reducing the amplitude of f_2 thus makes both amplitudes equal.

Observe that the outer peaks of modulation in Fig. 2C have sine wave shapes, but the valleys or troughs are V-shaped; this is clearly a case of amplitude distortion, for the modulation envelope no longer corresponds to the sine wave modulation signal (like that in Fig. 2A) at the input of the tuning circuit. (The curve in Fig. 2Cwas obtained by adding together and plotting the values of the carrier and the side frequency at each instant of time; it can be verified with a cathode ray oscilloscope and suitable special laboratory equipment.) Lower percentages of modulation than 100% and different off-tune carrier frequencies will of course give slightly different wave forms for the envelope in Fig. 2C, but amplitude distortion will be evident in all cases.

You can expect distortion similar to that in Fig. 2C whenever a tuning circuit is not properly tuned to the input carrier or when it has an unsymmetrical resonant curve, for both conditions result in unequal amplification of upper and lower side frequencies. The resulting amplitude distortion cannot be corrected for in the audio system; it will be present in the receiver output, and will often be annoying to the radio listener. The elimination of amplitude distortion in the R.F. system is therefore a matter of vital importance to the radio serviceman as well as the receiver designer.

Analyzing Typical Response Curves

Four typical resonant response curves of actual radio receivers, such as might be obtained by using a cathode ray oscilloscope and the necessary associated equipment, appear in Fig. 3. In each case f represents the carrier frequency to which the R.F. amplifier is tuned, while f_1 and f_2 represent the lowest and highest side frequencies involved. When properly interpreted, these curves reveal considerable information about distortion.

Sharp Peak. An R.F. amplifier having the sharp peak response curve shown in Fig. 3A will cause severe attenuation of the higher modulation frequencies (severe frequency distortion). (Remember that low modulation frequencies correspond to side frequencies close to and both above and below f, while high modulation frequencies correspond to side frequencies near f_1 and f_2 .) Since the gain at f_1 is less than at f_2 in this example, some amplitude distortion is also to be expected; this may not be severe, since the difference between the two values is not great. In R.F. amplifiers which have peak response curves, the greatest amount of amplitude distortion occurs because of improper tuning, which gives the condition represented by Figs. 2C, 2D and 2F. It is primarily for this reason that highly selective receivers, which naturally have sharply peaked response curves, are equipped with tuning aids.

Rounded Peak. Rounding or broadening of the peak of a response curve, by cutting down the amplification in the vicinity of the carrier frequency more than at the extreme side frequenthan the other, with resulting amplitude distortion. Furthermore, if the valley between the peaks is too deep, the lower modulation frequencies (having side frequencies in this valley) will be attenuated and frequency distortion will be evident.

Symmetrical Double Peak. When a serviceman adjusts a band-pass tuning circuit for high fidelity and good selectivity, his goal is the rarely-attained ideal square top response curve; ordinarily, however, he is entirely satisfied if he can secure the symmetrical double peak response curve shown in Fig. 3D, which has a negligible valley between the peaks. He knows that



FIG. 3. These four response curves are representative of those which can be viewed on the screen of a cathode ray oscilloscope when actual R.F. and I.F. amplifiers are being tested. The shaded areas and vertical lines are of course not seen on the C.R.O. screen; they have been added here in order to show the range of side frequencies handled by the amplifier along with the carrier in each case.

cies, is easily accomplished by a service technician who understands R.F. tuning circuits. The result is a broad peak response curve similar to that shown in Fig. *SB*, which gives considerably less frequency distortion at the expense of selectivity and gain.

Distorted Double Peak. Band-pass R.F. tuning circuits will, if properly designed and adjusted, give a double peak response curve with steep sides, insuring good fidelity and selectivity. Unless the adjustments are carefully made, however, there is a possibility that a distorted double peak response curve like that shown in Fig. SC, in which one peak is higher than the other, will be obtained. Naturally a curve such as this is undesirable, for one side frequency is amplified more when an R.F. amplifier has a symmetrical double-peak response curve such as this, amplitude distortion will not occur and frequency distortion will be negligible in the R.F. or I.F. amplifier. A properly designed band-pass tuning circuit can give far better selectivity than a circuit having a peak response which has been broadened to give equally as good fidelity. (The steeper the sides of the response curve outside the f_1 - f_2 region, the better is the selectivity.)

Alignment of R.F. Tuning Circuits. The final factory inspection of a radio receiver generally includes a check of the response curve for the R.F. section, to make sure that it has the desired shape. This is referred to as a check of the alignment. Oftentimes this alignment may be disturbed by rough handling during shipment and by general aging of the receiver, making it necessary for the serviceman to realign the tuning circuits.

It is a well-known fact that most radio receivers are designed to have a compromise between selectivity, gain and fidelity, so they will please the greatest possible number of listeners. It is when a particular listener wants the highest possible fidelity or wants maximum gain and selectivity for reception of distant stations that the serviceman is called in to change this compromise response characteristic. Realigning a receiver to have a sharppeak response gives maximum possible gain and selectivity; usually this is easily done with an ordinary all-wave which control the response characteristic.

Factors Controlling Response

The shape and height (maximum gain) of the response curve for a tuned R.F. amplifier are essentially determined by one or more of the following factors: 1, The Q factors of the coils used in the amplifier; 2, the L/Cratio of each tuned circuit in the amplifier; 3, the types of coupling used to connect the tuned circuits to each other and to vacuum tubes; 4, the characteristics of the vacuum tubes. Although these factors have been discussed to a certain extent in previous lessons, they are so important to our study of tuning circuits that I will review them briefly at this time.

Q Factor of a Coil. In previous les-



The response curve of the radio receiver at the right appears on the screen of the cathode ray tube in the radio servicing oscilloscope (center) when proper connections are made between the receiver, the oscilloscope and the frequency-wobbulated R.F. signal generator at the left.

signal generator and an output indicator. Correct aligning for high fidelity cannot be easily carried out without additional equipment, however; a frequency-wobbulated signal generator^{*} and a cathode ray oscilloscope are essential in this case. It is not the purpose of this lesson to describe the service procedures followed in realigning radio receivers, but rather to point out the various factors in tuned circuits

sons it was pointed out that any tuning circuit has a certain amount of loss due primarily to the A.C. resistance of the coil (the resistance of the condenser and the circuit wiring is so low that it is usually neglected entirely). The ohmic value of this A.C. resistance of a coil depends not only upon the D.C. resistance of the wire used in making the coil, but also upon "skin effects" associated with high frequency currents, upon losses occurring in the dielectric materials used for the coil form and insulation, and upon the nature of the load which is coupled to the coil. The Q factor of a coil was de-

^{*}A frequency-wobbulated signal generator is a special type of R.F. signal generator whose output frequency can be made to vary regularly and automatically above and below a definite R.F. value to cover any desired range of side frequencies.

Coil reactance in ohms Q factor of a coil= $\frac{\text{Coil} \text{ A.C. resistance in ohms}}{\text{Coil} \text{ A.C. resistance in ohms}}$

fined as the coil reactance divided by this coil A.C. resistance, all values being measured at the same frequency. Furthermore, since it is the coil which controls the tuning circuit losses, the Q factor of the coil can be considered as the Ω factor of the entire tuning circuit.

What Q Factor Tells Us About Tuning Circuits. The Q factor of the coil in a tuning circuit is a numerical value. often referred to simply as Q: it tells us the following important facts about the two types of tuning circuits: Series Resonant Circuits-

- 1. At resonance, the A.C. voltage across the coil is O times the source voltage.
- 2. At resonance, the impedance of the tuned circuit is entirely resistive, and is equal to the impedance of the coil in ohms divided by the Q factor of the coil.

Parallel Resonant Circuits--

- 1. At resonance, the current through the coil is O times the source current.
- 2. At resonance, the impedance of the tuned circuit is Q times the coil impedance, and is entirely resistive.

Up to a few years ago, engineers and scientists discussed the behavior of tuned circuits in terms of the A.C. resistance of the coil; this practice is quite correct, and may still be found in many text-books. Modern engineers

prefer to think in terms of Q factor. however, since they now have instruments with which they can measure the Q factor of a coil directly.*

One of the instruments available for measuring the Q factor of a coil is shown in Fig. 4; it is known as a Q factor meter, and can also be used for measuring the Q factor of any resistor or condenser

In a coil or condenser the radio engineer desires pure reactance at any frequency, with no resistance to cause loss of useful power, and consequently he wants the Q factors of these parts (the ratios of reactance to resistance)



Courtesy Boonton Radio Corp.

FIG. 4. This Q-factor meter is typical of the instru-ments used by radio engineers for measuring the Q factor of coils, condensers and resistors. The de-vice being measured (a coil in this case) is con-nected to two of the terminals at the top of the instrument.

to be as high as possible in most cases. The Q factor meter reveals that the Q factors of condensers are very high (resistance is very low) in comparison with coils: the meter can also be used to compare various coils (or condensers) as to quality.

Although a resistor is ordinarily thought of as a pure resistance, it often has appreciable inductance, particularly when of the wire-wound type. The Q factor of a resistor (the ratio of

^{*}O factor is actually the reciprocal of power factor. You will remember from previous Lessons that the power factor of a device is equal to its resistance divided by its impedance; where the impedance is essentially reactance, power factor can be considered equal to resistance divided by reactance, just as Q factor is equal to reactance divided by resistance under the same conditions. You can always find the Q factor of a device (if the power factor is known) by dividing the number 1 by the power factor. Good coils and condensers have a high Q factor and a low power factor : good resistors have a low Q factor and a high power factor.

Resistance. That opposition to current flow in an A.C. circuit which results in power loss; it is often called A.C. resistance.

Reactance. That opposition to current flow in an A.C. circuit_which does_not result in power losses. Reactance may be either inductive (due to a coil or inductance) or capacitive (due to a condenser or capacitance).

Impedance. The total opposition to current flow in an A.C. circuit. Impedance combines the effects of both resistance and reactance, and therefore determines how much alternating current will flow.

When the resistance of a device is very small with respect to its reactance, as in coils and condensers, the impedance will be just slightly larger than the reactance, and for all practical purposes we can consider the impedance and reactance to be equal.

reactance to resistance) should therefore be as low as possible in circuits where only resistance is desired.

How to Increase the Q Factor of a Coil. For a coil of given inductance, keeping the losses in the coil at a minimum insures a high Q factor. Losses which are due to capacity between the turns of the coil can be reduced by using insulating materials and coil coatings which have low dielectric losses. Losses due to the coil form can be reduced by improving the quality of the material used in the coil form. Losses due to skin effects in the wire at high frequencies can be reduced by using a large number of enamel-covered wires which are braided together to form what is known as "litz" wire. (Unfortunately, litz wire is valuable only at frequencies between about 200 and 900 kc.) At high frequencies, losses can be kept down by making the coil with large solid wire, with flat copper ribbon or with copper tubing, all turns being equally spaced.

The shape of a coil has considerable effect upon its Q factor, for coil losses vary with the shape of the coil, particularly when the winding is in several layers. When designing multilayer R.F. coils, radio engineers generally test out several shapes and select that which gives the highest Q factor for the coil.

Shielding an R.F. coil by placing it in a metal can or compartment increases the losses in the coil, reduces the coil inductance, and also reduces the coil Q factor. With a Q factor meter like that shown in Fig. 4, the designer can select a shield for a given coil which will not excessively reduce the Q factor.

Large coils which are made from heavy copper wire, tubing or ribbon can have Q factors of over 500. In radio receivers, where small coils must be used because of space limitations, Q factors of 150 are considered excellent; in order to obtain this high value for radio receiver coils, it is necessary to use coil forms at least two inches in diameter and use heavy solid wire or litz wire for the windings.

Even though an engineer is able to choose a coil size, shape and type of wire which will keep losses low and thus give a high Q factor at a particular frequency, this is no guarantee that the Q factor will remain high for other frequencies. The graph in Fig. 5 shows that the Q factor of a practical coil varies with frequency. This graph tells us that in general, a coil which has a very high Q factor at a low frequency will lose its Q factor rapidly at higher frequencies (curve 1), whereas a coil with a reasonably high Q factor at low and medium frequencies will tend to retain this Q factor value as frequency is increased (curve 2 in Fig. 5). Naturally these facts about how the Q factor for a coil varies with frequency are of extreme importance in connection with the tuning circuits of radio receivers, for these circuits are made to respond to a wide range of carrier frequencies.

L/C Ratio for Tuning Circuits. In a tuning circuit the ratio of coil inductance to condenser capacity.commonly known as the L/C ratio, oftentimes has an important effect upon the selectivity of the circuit, determining the ability of the circuit to reject frequencies which differ from the resonant frequency. In series resonant circuits a large L/C ratio (secured by using a high-inductance coil) gives best selectivity, but in parallel resonant circuits it is the lowest L/C ratio which gives the best selectivity (assuming, of course, that the losses in the coils are the same for all L/C ratios).

When comparing the L/C ratios for two different tuning circuits, it is important that the same unit of inductance be used for each coil and the same unit of capacity be used for each condenser. For example, a typical 500to-1.500 kc. tuning circuit uses a 250 microhenry coil and a 400 micro-microfarad (mmfd.) maximum capacity variable condenser; with this condenser set at 100 mmfd., the L/C ratio is $250 \div 100$, or 2.5. For comparison purposes, then, the capacity and inductance used in another circuit should be expressed in these same units when figuring out the L/C ratio.

Coupling Methods for R.F. Tuning Circuits

The method used for coupling an R.F. tuning circuit to a tube in an R.F. amplifier or for coupling two tuning

circuits together naturally has a great deal to do with the operation of the amplifier. Four basic coupling methods are in general use:

Method 1. Directly coupled resonant load like that in Fig. 6A, where a parallel resonant circuit is directly connected to the plate of the amplifier tube.

Method 2. Tuned secondary transformer load like that in Fig. 8A, where a sories resonant circuit is inductively coupled to the plate of the amplifier tube.

Method 3. Double-tuned transformer load like that in Fig. 9A, where two resonant circuits which are mutually coupled inductively serve as the plate load for the amplifier tube.

Method 4. Double-tuned capacity-coupled load, like that in Fig. 11A, where two resonant circuits which are mutually coupled capacitively serve as the amplifier tube plate load.

Coefficient of Coupling. We are particularly interested in the amount of coupling provided between the resonant load circuit and the plate circuit of the R.F. amplifier tube by each of the basic coupling methods. Direct coupling such as that in method 1 pro-



FIG. 5. Chart showing how Q factor varies with frequency for two representative broadcast band coils.

vides maximum possible coupling between the source (the vacuum tube plate circuit) and the load (the resonant circuit).

With transformer loads as in methods 2 and 3, where mutual inductance (M) provides the coupling between circuits, we can change the amount of coupling by changing the position of either L_1 or L_2 . When all of the flux produced by current flowing in the primary coil (L_1) links with the sec-

ondary coil (L_2) , we have maximum coupling and maximum possible mutual inductance. When the primary and secondary coils are so positioned with relation to each other that only a part of the flux produced by one coil links with the other coil, the mutual inductance between them will be low and we have the condition of weak coupling. Under this condition the ratio of the actual mutual inductance to the maximum obtainable mutual inductance with very close coupling is a measure of the amount of coupling; this ratio is always a number less than 1, and is known as the coefficient of coupling. When two resonant tuning circuits are coupled, as in methods 3 and 4, the coefficient of coupling is quite important in determining the response characteristic of the circuit.

Effect of Vacuum Tube Characteristics. At the present time the use of pentode tubes in the R.F. amplifiers of radio receivers is becoming almost universal. Screen grid tubes are still to be found in some receivers, but triode tubes will be found only in the very old R.F. amplifier circuits. There are two features of pentode tubes which are significant in connection with tuning circuits and which should therefore be considered before we study coupling methods: 1, pentode tubes have high A.C. plate resistance values, because the plate is farther away from the cathode than in an ordinary triode tube; 2, pentode tubes have high amplification factors, because the control grid is considerably closer to the cathode than in an ordinary triode tube. For example, a 6K7 super-control pentode tube has an A.C. plate resistance of about 1,-000.000 ohms and an amplification factor (μ) of about 1,000; a 6J7 pentode tube has an A.C. plate resistance greater than 1,500,000 ohms and a μ of over 1,500.

In a practical modern R.F. amplifier the effective load resistance rarely reaches a value higher than about onetenth the A.C. plate resistance of the pentode tube. This condition corresponds to that of a generator which has an internal resistance of at least 1.-000,000 ohms, connected to a load resistance-of not more than_100,000 ohms; you can readily see that variations in the load resistance will have little effect upon the A.C. plate current. You could actually short out the load resistance without affecting the A.C. plate current more than 10%, since the resistance of the generator itself has the greatest amount of control over circuit current. Engineers say that under this condition we have a constant-current generator.

Because the A.C. plate resistance of a pentode tube in an R.F. amplifier stage of a receiver is extremely high with relation to its plate load resistance, we can consider a pentode tube as a constant-current generator for any given grid input signal voltage, and thus simplify greatly our study of resonant circuits. The constant-value A.C. plate current which the tube will deliver is easy to determine if the mutual conductance of the tube is known. Here is the rule: The A.C. plate current in microamperes in a pentode tube is equal to the grid A.C. voltage in volts multiplied by the mutual conductance of the tube in micromhos. Remember that with pentode tubes, whatever changes you make in the tuning circuit or circuits which serve as the load will have only negligible effect upon the A.C. plate currents.

The high amplification factor of a pentode tube produces a reasonably high output voltage despite the great loss in signal voltage due to the fact that the load resistance is less than the A.C. plate resistance.

Importance of Studying Coupling Methods. Now let us make a more detailed study of each of the basic coupling methods in order to see what the engineer can do to secure the desired amount of selectivity, gain and fidelity for a particular purpose when designing radio apparatus using one of these circuits, and in order to see what the serviceman can do to alter the selectivity, gain or fidelity characteristics of a receiver. This discussion will show you the importance of making exact tube and parts replacements in R.F. circuits. Substituting some newly developed tube or part for older equipment may lead to serious trouble, unless the substitution is made with full knowledge grid input voltage e_g multiplied by the amplification factor (μ) of the tube; this is simply the equivalent vacuum tube circuit idea which you have already studied, and which can be proven correct either by mathematics or experiment. In the simplified circuit of Fig. 6B, this A.C. voltage e_p is divided between the load circuit (across points 1 and 2) and the A.C. plate resistance r_p ; only when the resonant resistance of this load is many times greater than r_p will the load get practically all of the source voltage e_p .*

GAIN





of the design problems involved in the change.

Directly Coupled Resonant Loads

Simplified Circuit. We can simplify our study of the directly coupled parallel resonant load circuit in Fig. 6A by omitting all parts which have no effect upon the performance of the amplifying circuit and redrawing our circuit in the form shown in Fig. 6B. Observe that the R.F. by-pass condensers, the D.C. voltage supply leads for the tube electrodes and the automatic C bias resistor have been omitted, and the resistance of coil Lis now represented by a separate resistor R. The vacuum tube has been replaced by an A.C. voltage source $e_{\rm p}$ (shown as an A.C. generator) in series with the A.C. plate resistance r_p , with the value of $e_{\mathbf{p}}$ being equal to the A.C.

FIG. 6B. Simplified equivalent circuit of the directly coupled resonant load arrangement in Fig. 6A.

FIG. 7. Effects of grid resistor values upon the shape of the response curve for the circuit of Fig. 6A.

Rg IS

REMOVED

Rg = NORMAL VALUE

Rg IS LOW IN VALUE

Effect of Coil Inductance and QFactor on Over-all Amplification. We know that the resonant resistance of the parallel resonant circuit in Fig. 6Bdepends upon the coil reactance and upon the Q factor of the coil; for any given frequency, then, increasing the inductance of the coil will increase its reactance, will increase the resonant load resistance, and will therefore increase the over-all amplification. Likewise, increasing the Q factor of the coil will increase the over-all amplification.

^{*}The voltage produced across the load for a one-volt grid input signal is a measure of the true amplification of a stage. In any pentode circuit such as we have here, we simply multiply the mutual conductance of the tube in micromhos by the resonant resistance of the load in ohms and divide the result hy 1,000,000 to get the true or overall amplification. Any change which increases the load resistance will therefore increase the over-all amplification.

For circuits which handle only a single frequency, such as I.F. amplifier stages, the designer endeavors to select a coil which has the highest usable inductance and at the same time has a high Q factor. From this it should be obvious to you that when a coil in the tuning circuit of an R.F. amplifier becomes defective, the mere substitution of another coil having the same inductance is not a guarantee that the correct over-all amplification will be obtained. The practical radio man uses exact duplicate replacement coils in order to make sure that he is using a coil with the correct Q factor as well as the correct inductance.

Effect of Frequency upon Over-all Amplification. Suppose we have a parallel resonant load circuit which tunes over the frequency range from 500 kc. to 1,500 kc.; will the over-all amplification remain the same at all frequencies in this range? Curve 2 in Fig. 5 shows that as frequency is increased above the middle-frequency range, the Q factor of a practical radio. coil decreases: this reduction in Q factor would tend to reduce the overall amplification at the higher frequencies. On the other hand, increasing the frequency three times (from 500 kc. to 1,500 kc.) would increase the reactance of the coil three times, thus increasing the resonant load resistance three times and consequently increasing the over-all amplification about three times. In most cases this increase in amplification due to an increase in coil reactance will completely overshadow the decrease due to a reduction in Q factor at high frequencies. Consequently we can say that increasing the signal frequency being fed to a tuned R.F. amplifier which uses a parallel resonant circuit as a plate load will in most cases increase the over-all amplification of the amplifier. Only when the Q factor drops rapidly with frequency, as in curve 1 in Fig. 5, will

the over-all gain remain constant or drop when frequency is increased.

Effect of Frequency upon Selectivity. The selectivity of an R.F. amplifier circuit is defined as the ratio of the amplification provided at the desired signal frequency to the amplification provided at the nearest undesired signal frequency. The effect of frequency upon selectivity can be demonstrated by considering an actual case, that where a broadcast band receiver using the circuit represented by Figs. 6A and 6B is first tuned to 500 kc. and then to 1,500 kc. For simplicity we will assume that in each case the nearest undesired frequency is 100 kc. away.

Remember that lowering the resonant resistance (or reactance if off resonance) of the tuned load circuit in this amplifier will lower the over-all amplification of the amplifier. Let us first assume that the Q factor remains constant over the range from 500 kc. to 1,500 kc.* Under this condition we know that the amplification at resonance for 1,500 kc. will be three times the amplification for 500 kc.

At frequencies considerably off resonance, a parallel resonant circuit like that in Fig. 6B acts like that re-(capacitive or inductive) actance which is lowest in ohmic value. At 400 kc. (100 kc. below the desired 500 kc. signal), then, the circuit will have a reactance essentially equal to the reactance of the coil, and this reactance will determine the amplification at this nearest undesired signal frequency in our example. Selectivity at 500 kc. will then be the ratio of the amplification at 500 kc. to the amplification at 400 kc. At 1,400 kc. (100 kc. below the desired 1,500 kc. signal), the reactance of the coil will be approximately three times its value at 400 kc. and there-

^{*}The Q factor of a practical coil actually varies considerably with frequency; a constant Q factor is assumed here in order to simplify this discussion.

fore the amplification at the nearest undesired signal in this case will be three times what it was for the nearest undesired signal in the previous case. Since the amplification at 1,500 kc. is likewise three times the value at 500 kc., we will get the same selectivity ratio in both cases. This means that when the Q factor is assumed to be constant over the tuning range, the selectivity of a parallel resonant circuit will likewise remain essentially constant over the tuning range.

With the coils generally used in the tuning circuits of radio receivers, however, the Q factor will be found to decrease considerably as frequency is increased. This decrease in Q factor lowers the amplification at resonance but has no effect upon amplification at off-resonance frequencies; consequently the selectivity ratios for higher frequencies will be reduced. In actual circuits the selectivity will vary in much the same manner as the Q factor of the coil varies: the curves in Fig. δ thus can tell us how selectivity varies with frequency. In general, you will find it easier to separate stations in the middle region of a radio receiver tuning range than at the extreme high or low frequency ends, for coil Q factors are generally highest in the middle region.

Effect of C Bias Voltage Variations. It is a known fact that in super-control pentode tubes, increasing the negative C bias voltage has the effect of decreasing the mutual conductance of the tube, thereby lowering the A.C. plate current and reducing the over-all amplification of the stage. Variations in C bias voltage have little effect upon the selectivity of R.F. amplifiers using pentode tubes, however, for the resonant load characteristics are not affected by C bias voltage variations.

Effect of Loading the Tuned Circuit. There is one simple way of changing the selectivity and gain of a tuned circuit such as that shown in Fig. 6A, and this is to change the load on the tuned circuit by changing the value of grid resistor $R_{\rm g}$. Obviously the gain will be lowered when the value of $R_{\rm g}$ is reduced, for this grid resistor acts in parallel with the resonant circuit and therefore reduces the load resistance in the plate circuit of the tube (assuming that coupling condenser $C_{\rm K}$ has negligible reactance). Off resonance, however, the value of $R_{\rm g}$ has little effect upon the amplification of undesired



All-wave superheterodyne receiver being aligned for peak response. An R.F. signal generator and an output meter are the only instruments needed. The output of the R.F. signal generator section is fed into the receiver, and the multimeter section is used as n output meter.

frequencies, for now the reactance of the tuned circuit will be considerably lower than the value of R_s ordinarily used, and the reactance of the tuned circuit will control amplification Loading the tuned circuit of an R.F. amplifier by reducing the value of R_s thus *lowers selectivity* by lowering the amplification ratio for desired and undesired signals.

The effects of various values of $R_{\rm g}$ are shown graphically in Fig. 7; the middle response curve is for the condition where the usual fairly high value of $R_{\rm g}$ is in the circuit of Fig. ∂A . The lower dotted curve is for a lower value of $R_{\rm g}$, and clearly shows that both the

gain and the selectivity of a tuned circuit in an R.F. amplifier are lowered when the tuned circuit is loaded by reducing the ohmic value of the grid resistor for the following stage; fidelity is considerably improved, however, for the broad peak insures uniform amplification of all side frequencies. The uppermost dotted curve is for the condition where R_{g} is removed entirely; now amplification is very high and selectivity is good, but fidelity is very poor because side frequencies are amplified very much less than the carrier frequency. Thus you can see that the fidelity of a receiver using a tuning circuit like that in Fig. 6A could be improved by reducing the value of $R_{\rm s}$, provided that a loss in amplification and selectivity is permissible. Likewise, DX (distance-getting) performance could be increased by using a higher value of R_{π} .

High-gain Directly Coupled Resonant Load Circuit. In some inexpensive receivers which have few tubes, the circuit in Fig. 6A is modified slightly to eliminate the amplification-reducing effect of $R_{\rm g}$. Instead of coupling to the next tube through C_{κ} and R_{κ} , a second coil is inductively coupled to coil L and connected to the grid and cathode of the following tube. (The circuit arrangement is exactly as in Fig. 9A with tuning condenser C_2 removed.) By using a large mutual inductance between the two similar coils, the entire resonant circuit voltage can be transferred to the following stage without appreciable loss. If there are more turns on the secondary than on the primary, a step-up in output voltage can be secured. Fidelity is somewhat poor with this arrangement, however, for a sharp peak response curve is secured. The peak can be broadened by shunting the tuning condenser (C in Fig. 6A) with a 20,000 to 200,000 ohm resistor, but this will, of course, lower the gain.

Tuned Secondary Transformer Loads

The R.F. amplifier circuit arrangement shown in Fig. 8A, which uses a tuned secondary transformer_load, is widely used in tuned radio frequency receivers and in the station selector (preselector) circuits of superheterodyne receivers. Since it is modern practice to use pentode or screen grid tubes in this circuit also, the A.C. plate current will be essentially independent of conditions in the resonant load circuit; we will assume this condition during our discussion of this circuit.

A simplified version of the tuned secondary transformer load circuit in Fig. 8A appears in Fig. 8B. Observe



FIG. 8A. Tuned secondary transformer load circuit as commonly used in a pentode R.F. amplifier stage. 1

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that the primary and secondary coils, L_1 and L_2 , are each divided into two parts in this equivalent circuit. Those sections on each coil which link each other completely through mutual inductance M provide the only coupling between the two circuits; the remaining sections, which do not link each other at all, are known as the primary and secondary leakage inductances respectively. The primary leakage inductance is always equal to the original primary inductance minus that inductance which totally links with the secondary coil, and this totally-linking portion is in turn equal to the primary inductance multiplied by the coefficient of coupling of the original circuit. This same reasoning also applies to the sections of the secondary inductance.

Resonance exists in the secondary circuit for a desired signal frequency when the reactance of tuning condenser C_2 is equal to the reactance of the secondary leakage inductance at that frequency; the reactance of the other secondary coil section can be neglected, for it is cancelled out through mutual inductance M by the corresponding primary coil section. At resonance, then, the secondary resistance R_2 is the only factor which limits secondary current. This secondary resistance has an effect upon the primary circuit; engineers say that it is reflected into the primary circuit, with the reflected value of resistance being determined by the value of mutual inductance M, by the signal frequency and by the original value of R_2 . Increasing the mutual inductance, increasing the frequency or decreasing the ohmic value of R_2 will increase the value of reflected resistance in the primary circuit." When the ohmic value of the reflected resistance equals the A.C. plate resistance of the tube, maximum gain is obtained in this tuned secondary transformer load circuit. It is almost impossible to secure this condition with screen grid and pentode tubes because of their high A.C. plate resistance values, but it can be done with triode tubes which have low A.C. plate resistance values.

In all practical circuits which use screen grid or pentode tubes, the reflected resistance in the primary circuit is negligibly small in comparison to the A.C. plate resistance of the tube. The primary signal current is therefore unaffected by any conditions in the secondary circuit which might change the value of reflected resistance. Furthermore, the reactance of the primary leakage inductance is also negligibly small in comparison to the A.C. plate resistance, and consequently the changes in this reactance with frequency can be neglected. Thus we find that the only two factors which control the A.C. plate current in a practical R.F. amplifier of the tuned secondary transformer load type are the mutual conductance of the tube and the applied A.C. grid voltage. The signal voltage e, which is induced in the secondary depends upon this A.C. plate current and mutual reactance of M.



FIG. 8B. Simplified equivalent circuit of the tuned secondary transformer load circuit in Fig. 8A.

This means that desired as well as undesired signal voltages applied to the grid of the tube in Fig. 8A will produce the same induced voltage in the secondary circuit of the plate load; it remains for the series resonant secondary circuit to tune out all but the desired frequencies.

Effect of Tuned Secondary Circuit on Amplification. The voltage across secondary tuning condenser C_2 in Fig. 8A is applied directly to the grid of the following tube, and therefore anything which increases the value of this voltage at resonance will make the over-all amplification of the stage greater. Since the tuned secondary is a series resonant circuit, this voltage is equal to the source voltage e_8 multiplied by the Q factor of the secondary coil. Anything which increases the Q factor of the coil therefore increases

^{*}Although undoubtedly you will never have to determine exactly what the reflected resistance value is, the formula for doing this is presented here for reference purposes: Multiplying the mutual reactance of M by itself once and then dividing by the secondary circuit resistance R_s gives the reflected resistance in the primary circuit. (Reflected resistance in ohms = $2\pi f M \propto 2\pi f M + R_s$, where $\pi = 3.14$, f = frequency in cycles and M = mutual inductance in henrys.)

the over-all amplification. This is a good reason for using high Q coils in circuits of this type.

For a given grid input voltage, increasing the secondary induced voltage will also inorcase the over-all amplification of the stage; this can be done by using a tube which has a higher mutual conductance, by increasing the mutual inductance M between the primary and secondary, and by increasing the frequency of the desired signal.

To reduce the gain of an R.F. amplifier circuit like this, as is often necessary in actual receivers in order to prevent overloading of one or more following stages or to reduce the output volume to a desired lower level, the mutual conductance of the tube can be reduced. In a super-control pentode tube, the usual procedure for doing this involves increasing the negative C bias voltage on the tube.

When tuning the secondary circuit over a given frequency range, the amplification of the circuit will depend upon the manner in which the Q factor of the coil varies with frequency. Q factor normally decreases with frequency, but the reduction in gain due to this effect will be offset by an increase in gain due to the increased mutual reactance* of M at higher frequencies; the result is that the amplification of a typical tuned secondary transformer load circuit varies slightly over its tuning range.

Effect of Tuning Circuit on Selectivity. If the Q factor of the coil in a tuned secondary transformer load circuit remained constant over a given frequency range, the selectivity would also remain constant; Q factor decreases at the higher frequencies, however, so selectivity will likewise decrease. The peak of the response curve can be broadened or rounded by shunting the tuning condenser in Fig. 8A with a 20,000 to 100,000 ohm resistor. This will reduce the gain at resonance by reducing circuit current but will have little effect at off-resonance frequencies.

Double-Tuned Transformer Loads

The double-tuned transformer load circuit shown in Fig. 9A is widely used in the I.F. stages of superheterodyne receivers. One advantage of this circuit is that it can provide high selectivity while keeping the number of tubes at a minimum; another advantage is that the circuit can be adjusted to give an almost flat-top response curve for high fidelity. We will consider first the adjustment of this circuit for peak response and for band-pass response, and will then analyze the factors whick





control the circuit gain, selectivity and fidelity.

The circuit of Fig. 9A has been redrawn in simplified form in Fig. 9B in order that we can concentrate our study upon those parts which affect the performance of the tuned circuit. Again we have the coils divided into totally coupled sections and leakage inductance sections, as before. R_1 represents the A.C. resistance of the primary coil, while R_2 represents the A.C. resistance of the secondary coil.

Adjusting for Peak Response. With the circuit of Fig. 9B, a single-peak response characteristic can be obtained at any desired I.F. value by adjusting tuning condenser C_2 until its reactance

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^{*}Mutual reactance equals mutual inductance times frequency in cycles per second times the number 6.28.

exactly equals the secondary leakage reactance of the secondary winding at that I.F. value. Condenser C_1 is then adjusted in the same way in order to tune the primary circuit to resonance at the desired I.F. value. In a practical case this adjustment is made by connecting an R.F. voltmeter across C_2 to measure the secondary circuit output voltage E_2 ; C_1 and C_2 are then adjusted for maximum voltmeter reading.

At resonance, there is only the secondary circuit resistance R_2 to be reflected into the primary tuned circuit through mutual inductance M. This increases the resistance in the primary circuit, and therefore decreases the Q factor of the primary circuit. The presence of the secondary circuit thus reduces the voltage across primary tuning condenser C_1 , thereby reducing the amount of resonant stepped-up current through coil L_1 and reducing the amount of voltage induced in the secondary winding for resonance step-up by the secondary series resonant circuit. A double-tuned transformer load circuit which uses identical coils in both tuning circuits always gives less gain than a single parallel resonant load circuit using only one of these coils.

Increasing the mutual inductance Mby increasing the coupling between primary and secondary coils tends to make the resistance which is reflected into the primary circuit larger, thus reducing E_1 , but at the same time the increased mutual inductance serves to increase the voltage which is induced in the secondary. Since the two effects tend to offset each other, there is naturally a particular value of mutual inductance which will give the highest possible gain. By experiments as well as calculation, engineers have determined that this optimum condition occurs when the mutual inductance Mis such that the resistance which is reflected into the primary circuit is exactly equal to the primary circuit resistance R_1 .

In practical radio circuits, condensers C_1 and C_2 in Fig. 9A are usually of the same capacity, and consequently the primary and secondary coils must be alike in order to secure resonance at the same frequency. Since doubletuned transformer load circuits are ordinarily found in I.F. amplifier stages, these condensers will be of the trimmer type, independently adjustable. With any given coils, the condenser settings required for resonance are definitely fixed, and only the coupling between coils can be varied in



FIG. 9B. Simplified equivalent circuit of the doubletuned transformer load circuit in Fig. 9A.

order to secure optimum conditions. That coupling which gives the maximum possible gain is called the optimum or critical coupling. With identical coils, optimum coupling is obtained when the coefficient of coupling is exactly equal to 1 divided by the Q factor of the coil. For example, if the coil has a Q factor of 100, the coefficient of coupling for optimum gain will be $1 \div 100$, or .01.

Once a double-tuned transformer load circuit is adjusted for optimum coupling, either increasing or decreasing the coupling from this value will reduce the over-all gain. Increasing the coefficient of coupling also reduces the selectivity, broadening the peak of the response curve because of the increase in the resistance reflected into the primary circuit, but decreasing the coupling serves to increase the selectivity. When double-tuned transformer load circuits are used to give a single-peak response, the coupling is kept less than the critical value (the coils are undercoupled) in order to improve the selectivity at a sacrifice of gain.

Adjusting for Double-Peak Response. Let us assume first that there is no coupling whatsoever between the secondary circuit and the primary circuit, both being tuned to the frequency of the source. Clearly there will be no voltage induced in the secondary coil under this condition, and consequently there will be no output voltage. Now as we bring the secondary coil closer to the primary coil, energy will be transferred through the mutual inductance



FIG. 10. Effect of variations in soupling upon the shape of the response curve for the double-tuned transformer load circuit in Fig. 9A.

between the two coils and an output voltage will be produced across C_2 . As we gradually increase the coupling up to the critical coupling value, this output voltage will continue to increase. Increasing the coupling beyond the critical value will at first have no effect upon the output voltage and will then gradually cause the output voltage to lower. What actually happens is shown in Fig. 10; curve 1 represents a very small amount of coupling, while curves 2, 3, 4 and 5 represent increasing greater amounts of couplings, with curve 3 representing the conditions for critical coupling.

Let us see why double peaks occur in curves 4 and 5 in Fig. 10. When coupling is below the critical value, both primary and secondary circuits have the same resonant frequency (assuming correct tuning) and single-peak response is secured. Increasing the coupling beyond the critical value without changing the tuning condenser settings causes the leakage inductance in each circuit to decrease, and consequently the resonant frequency of the secondary circuit becomes higher than before (the lower the effective inductance for a given capacity value, the higher is the resonant frequency of a resonant circuit). Increasing the coupling beyond the critical value likewise lowers the primary leakage inductance, making the primary circuit resonate also at a higher frequency. A signal at this higher frequency will thus undergo resonant step-up in both the primary and secondary circuits, giving a high output voltage.

When we exceed the critical coupling value, we have another interaction between the two circuits to consider. Consider conditions for a signal which is lower than the original resonant frequency of the secondary. At this lower frequency the series resonant secondary circuit will act as a capacity and will reflect into the primary circuit as an inductance which increases effective primary inductance, the bringing the primary circuit to resonance at this lower frequency. Likewise, the parallel resonant primary circuit alone will act as an inductance at the lower frequency and will reflect into the secondary as a capacity which brings the secondary to resonance at the lower frequency also. Primary and secondary circuits are thus resonant to both a higher and a lower frequency than the resonant frequency of either circuit alone, and as a result we have double-peak or band-pass response.

At critical coupling, an essentially flat-top response curve is obtained, with fairly steep sides and double peaks just beginning to form. The ap-

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RESONANT CIRCUIT DATA

At frequencies below the resonant frequency:

A series resonant circuit acts as a capacity.

A parallel resonant circuit acts as an inductance.

At frequencies above the resonant frequency:

A series resonant circuit acts as an inductance.

A parallel resonant circuit acts as a capacity.

General rule: At any off-resonant frequency, a series resonant circuit acts like that part which has the highest reactance, while a parallel resonant circuit acts like that part which has the lowest reactance.

proximate distance between these peaks, as measured in kc., is easily computed: it is equal to the coefficient of coupling multiplied by the carrier frequency to which the R.F. amplifier is tuned. For example, suppose that the coils in Fig. 9A each have a Q factor of 150, and the two resonant circuits are tuned to 460 kc. Critical coupling will be obtained when the coefficient of coupling is 1 divided by 150, which equals .0067; the separation between peaks at this condition of critical coupling will then be equal to 460 (the frequency in kc.) multiplied by .0067, or about 3 kc. The actual practical band width for this particular case will be somewhat wider (about 4.5 kc. in this example), for it is determined by the distance between the steep sides of the response curve rather than by the distance between the peaks.

In the above example, we can triple the amount of separation between peaks (thereby tripling the bandwidth) by tripling the amount of coupling, but this will give a valley between peaks as in curves 4 and 5 in Fig. 10. If, in addition to tripling the coupling, we reduce the Q factors of the coils to one-third of their original values, we can secure this same tripling of bandwidth without having a valley between peaks, but the gain will be considerably less when this is done. Clearly we must sacrifice one advantage in order to secure another, in this particular case.

When an engineer designs band-pass circuits like that in Fig. 9A, he endeavors to choose coils which will give the desired band-width at critical



Courteey Aladdin Radio Industries, Inc.

This Aladdin type D I.F. transformer provides any desired performance characteristics from high fidelity to extreme selectivity without appreciable variation in gain, at the option of the user. Rotating the center coil by means of a knob on the front panel of the receiver serves to vary the coupling between the other two coils. In the under-coupled position, as represented by surve A, we have sharp tuning and severe cutting of side frequencies above about 4,000 cycles, while curve B corresponds to overcoupling, with practically uniform amplification of all side frequencies up to 10,000 cycles off resonance. Three built-in trimmer condensers, one for tuning each coil, give additional control over transformer characteristics.

coupling. If he finds that this is impossible, he increases the coupling beyond the critical value enough to secure the desired band-width. This gives somewhat lower gain (curve δ in

Fig. 10 represents lower gain than curves 3 and 4).

Adjusting Actual Band-Pass Circuits. Band-pass circuits in practical radio frequency amplifiers are easily adjusted with the aid of a cathode ray oscilloscope which is connected to reproduce the response curve of the amplifier. By watching the effects of each adjustment upon the shape of this response curve, the radio serviceman knows definitely when he has secured the desired shape.

Ordinarily a serviceman does not know whether a particular band-pass circuit is over-coupled (greater than critical coupling), or is under-coupled (less than critical coupling). For this reason he must always try to adjust the circuit for peak response before making band-pass adjustments. In the case of Fig. 9A, this is done simply by feeding into the amplifier an R.F. signal of a definite frequency and adjusting tuning condensers C_1 and C_2 for maximum output voltage at the detector load, as indicated either by an output meter or by a cathode ray oscilloscope. It does not matter which condenser is adjusted first. The adjustments are repeated several times. If there are several band-pass circuits in an amplifier, each is adjusted in this same way for peak response. Inability to secure a single-peak response when this is done means that the coils are over-coupled; in this case the preliminary adjustment is omitted.

A cathode ray oscilloscope and a variable frequency signal generator are now connected to the amplifier in the proper manner to produce on the cathode ray oscilloscope screen the actual response curve of the entire amplifier. Assuming that the circuit in Fig. 9A is the only band-pass circuit in the amplifier, the capacity of C_1 is increased slightly and the capacity of C_2 is decreased the same amount (or C_1 may be decreased and C_2 increased) to make the flat top appear. If the coils have approximately critical coupling, only small changes in the condenser settings will be needed to secure double peaks. When coupling is considerably less than the critical value, however, the two condensers may have to be changed considerably from their peak response settings before the two peaks will appear. If the coils in a double-



Courtesy Trislett Biectrical Instrument Co. A cathode ray oscilloscope (lower section) and a trequency-modulated signal generator (upper section) are here mounted in a single cabinet. They can be connected to a radio receiver in a few minutes for producing the response curve of the receiver, and have a host of other radio servicing and testing uses as well.

tuned circuit are over-coupled, it will be impossible to adjust for a singlepeak response; the two peaks will always be present, and changes in condenser settings will merely serve to change the distance between peaks and alter the symmetry of appearance.

The more the condenser settings are changed, the greater will be the distance between the two peaks and the deeper will be the valley between the **peaks**. If the peaks of the response



curve of a double-tuned circuit are excessively high with respect to the valley between them when the desired band-width is secured, they can be reduced by shunting the primary and secondary circuit tuning condensers with 20,000 to 100,000-ohm resistors. This actually pulls the two peaks closer to the level of the valley, thus making the response more uniform over the entire band-width. It may be necessary to experiment with different values of resistors in the primary and secondary circuits in order to reduce both peaks equal amounts, if condenser adjustments alone are not sufficient to give a symmetrical response curve.



FIG. 11A. Double-tuned capacity-coupled load circuit as commonly used in a pentode R.F. amplifier stage.

In many modern high-fidelity receivers, controls are provided which vary the coupling between the primary and secondary coils of one or more double-tuned transformer load circuits in order to permit a choice between peak response and band-pass response. These transformers will have critical coupling when this control is set for band-pass performance. In some circuits which use variable coupling, you may find that the variable coupling control is labeled "volume control" on the schematic circuit diagram and on the panel of the receiver. In this case the coupling will always be less than the critical value, and under this condition any changes in coupling will affect the receiver gain far more than it will the receiver selectivity. We thus have simply a unique volume control.

Double-Tuned Capacity-Coupled Loads

Two resonant circuits may be coupled together by capacity coupling, as illustrated in Fig. 11A, instead of inductive coupling. Circuit action is much the same with either type of coupling, so there should be no difficulty in understanding how the simplified version of this capacity-coupled arrangement, shown in Fig. 11B, behaves under various operating conditions.

When both tuned circuits are at resonance, the secondary tuned circuit made up of L_2 , C_2 and C_M will act as a resistance. If we measure the resonant resistance between points \mathcal{B} and \mathcal{S} , in



FIG. 11B. Simplified equivalent circuit of the double-tuned capacity-coupled load circuit in Fig. 11A.

the circuit of Fig. 11A, we find it to be quite high.

You will note that tuning condenser C_2 and coupling condenser C_M are in series between points 2 and 3. A part of the total high resistance between these points will therefore exist across the terminals of coupling condenser C_M , and experience has shown that the actual value of this resistance will be proportional to the reactance of C_M . Increasing the reactance of coupling condenser C_M will therefore increase the effective resistance across it and thereby increase the resistance which is reflected into the primary circuit.

At resonance, the resonant resistance of the primary tuning circuit made up of L_1 , C_1 and C_M determines the value of the signal voltage E_1 which will be produced across the primary coil.
Again we assume that the pentode tube maintains the plate current $i_{\mathbf{p}}$ essentially constant; the resonant circuit current through the primary coil is of course greater than $i_{\rm p}$ and varies in value as circuit conditions are changed. The greater the resistance reflected into the primary circuit by the secondary, the lower will be the resonant circuit current through the primary coil and the lower will be the voltage across the primary. Just as with inductive coupling, maximum secondary circuit output voltage is obtained with critical coupling between the primary and the secondary, under which condition the resistance reflected into the primary is equal to the primary circuit resistance R_1 . When C_1 is equal to C_2 , as is usually the case in a practical circuit, the coefficient of coupling is the ratio of C_1 to C_M .

Just as with other band-pass circuits, increasing the coefficient of coupling beyond the critical value results in a double-peak response curve. The reasons for the existence of a double peak under these conditions are quite easily understood. When the capacity of coupling condenser C_{M} is made lower than the critical value, thereby raising its reactance, raising the voltage developed across it for transfer from primary to secondary, and increasing the coupling, the combined capacity of condensers C_2 and C_M in series will be less than before, making the resonant frequency of the secondary higher than before. Likewise the lower capacity of $C_{\rm M}$ in series with C_1 will raise the resonant frequency of the primary to the same high value. We have thus accounted for the higher-frequency peak on the band-pass response curve.

Now consider the effects of interaction between the two tuned circuits when the coefficient of coupling is beyond the critical value and the incoming signal is below the resonant frequency. Looking at the primary cir-

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cuit first, we note that secondary circuit components L_2 and C_2 in series are shunted across C_M ; the reactance of C_2 being greater than that of L_2 , this shunt combination will act as a condenser in parallel with C_M , increasing the effective capacity of C_M in the primary circuit. This higher value of C_M acting in series with C_1 increases the effective capacity in the primary circuit and therefore causes the primary



Courteey Aladdin Radio Industries, Inc.

Here is an adjustable coupling I.F. transformer (Type A Aladdin Polyiron transformer) which you may encounter in high quality radio receivers and particularly in commercial receivers. The curves at the right show its performance at different degrees of coupling. Note that the coils are at right angles to each other; moving the lower coil in either direction along its shaft by means of adjusting screws changes the coupling. Critical or optimum coupling is secured when the lower coil is not directly under the upper coil; curve B represents this conditioo, giving the ratio of the voltage output at resonance to the voltage output at frequencies up to 40 kc. above and below resonance. Amplification at the resonant frequency (465 kc.) is 250 for this condition, an unusually high value. Moving the coil more nearly under the upper coil gives undercoupling, with reduced gain but improved selectivity, as indicated by curve A. Moving the lower coil out from the critical coupling position gives overcoupling, with double peak response for high-fidelity band-pass results.

circuit to resonate at a lower frequency than the original value. Exactly the same analysis will show that the secondary circuit will also resonate at this lower frequency. Thus we have accounted for the lower-than-normal peak in the band-pass response curve. The reason why double peaks are not secured when the coefficient of coupling is less than the critical value is quite simple. Under this condition the capacity of $C_{\rm M}$ is so high that interaction between the two circuits is negligible. As a result, each circuit resonates at only one frequency, and a single peak response curve is secured.

With capacitive coupling between two resonant circuits. the separation between peaks at critical coupling is equal to the operating frequency multiplied by the coefficient of coupling. just as in the case of inductive coupling. Increasing the Q factor of the coils in the circuit of Fig. 11A increases the value of coupling required for critical coupling, thereby lessening the separation between peaks at critical coupling, but higher gain is secured. Increasing the coupling (by reducing the capacity of $C_{\rm M}$) increases the separation between peaks and therefore increases the band-width.

Just as with any double-tuned circuit, the separation between the two peaks may be increased, giving greater band-width, by detuning the primary and secondary circuits (by increasing the capacity in one circuit and decreasing the capacity an equal amount in the other). This deepens the valley between the peaks, but loading of the primary and secondary circuits with resistors will flatten the peaks and give a more uniform response curve over the entire band-width.

The procedure just described for widening the band-width and loading the tuning circuit is commonly used in the tuned circuits of television receivers, where a band-width of the order of 6 megacycles (6,000,000 cycles) is generally required. This procedure reduces the gain considerably, and consequently television receivers require more amplifier stages than broadcast band receivers. Both capacitive and inductive coupling are used in the double-tuned circuits of television receivers.

Combination Capacitive and Inductive Coupling

In tuned circuits which are to operate over a definite range of radio frequencies, such as the preselector circuits in superheterodyne receivers, the amount of coupling between tuned circuits is not entirely independent of frequency. As a result, the response curve of the circuit will vary in shape and

Fixed mics condensers are permanently connected across the coils in this type L inductance-tuned Aladdin I.F. transformer. Tuning is accomplished by changing the positions of the Polyiron pulverized iron cores inside each coil; this is done by adjusting the set screws at the top, labeled *PRI*. and *SEC*. Coupling is always less than the critical value. A gain of about 100 is secured when the coils are adjusted for peak contended to the coils are set.



Courtery Aladdin Radio Industries, Inc.

size at different frequencies, with the nature of the variation being dependent upon the type of coupling used.

Consider the double-tuned transformer load circuit in Fig. 9A first. Assuming that the coil Q factors remain the same as we tune from a low to a high radio frequency, we see that the reactance of mutual inductance Mincreases with frequency, producing a greater induced voltage in the secondary circuit and therefore giving a greater gain at the higher frequencies.

With the capacity-coupled circuit shown in Fig. 11A, on the other hand, the reactance of coupling condenser $C_{\mathbf{M}}$ decreases as we tune from a low to a high frequency, with the result that less voltage is fed into the secondary tuned circuit at higher frequencies, and less gain is realized at higher frequencies.

Capacitive and inductive coupling are often used in the same circuit in order to make an amplifier have the same gain at both low and high frequencies. A typical band-pass R.F. amplifier circuit which uses both types of coupling is given in Fig. 12; as you can see, primary coil L_1 is inductively coupled (through mutual inductance M) to secondary coil L_2 , and these two coils are also coupled together capacitively by coupling condenser C_{M} . The antenna is inductively coupled to coil L_1 in the first tuned circuit through mutual inductance M_{A} . Resistor R_{a} . having a high ohmic value, provides



FIG. 12. Double- tuned R.F. amplifier input circuit using both inductive (M) and capacitive (C_M) coupling to equalize the gain at all frequencies in the tuning range.

a conductive path around coupling condenser C_{M} , so that the negative C bias voltage across R_{C} and C_{C} will be applied to the grid of the tube.

Tuned Secondary with Fixed Tuned Primary Circuit

When a single resonant circuit gives more gain at high frequencies than at low frequencies, the circuit illustrated in Fig. 13 is often used to equalize the gain over the entire tuning range. The primary winding L_1 has a distributed capacity between turns which is in effect equivalent to a condenser C_1 connected across the winding. The primary coil can be so designed that the coil and its distributed capacity form a parallel resonant circuit which resonates at a low frequency in the tuning range, giving resonant current step-up at low frequencies and thereby inducing larger voltages than normal in the secondary circuit at the lower frequencies in the tuning range. At the higher frequencies, however, the primary circuit is off resonance and the value of mutual inductance *M* alone determines the amount of voltage induced in the secondary.

Unfortunately, this use of a fixed tuned primary circuit to boost the gain at low frequencies works entirely too well, boosting the gain at low frequencies so much that we have the reverse of the initial unequal gain condition. It is for this reason that a small amount



FIG. 13. Another R.F. amplifier input circuit which uses both inductive and capacitive coupling to equalize the gain over the tuning range. Fixed tuning of the primary coil is provided by the distributed capacity (C_1) between turns of the coil.

of capacity coupling is used between the high R.F. terminal of the primary coil and the high R.F. or grid terminal of the secondary coil; a stiff copper wire connected to the primary and curled partly around the secondary winding at the grid terminal end provides sufficient capacity coupling for the purpose. This wire does not connect to the secondary coil, as only capacity coupling is desired.

The circuit diagram in Fig. 13 illustrates the use of this capacity-coupling wire in the antenna system of a receiver. Antenna coil L_1 , along with its distributed capacity, tunes to about the lowest frequency in the tuning range of the secondary circuit (this tuning range being controlled by L_2









Typical R.F. transformers which can be used in the tuning circuits described in this lesson. Getting the required inductance is only a small part of the coil designer's work; he must also consider such important things as the Q factor and how it varies with frequency, the degree of coupling needed for desired results, effect of the shield, and gain-equalizing methods. The examples shown here are all Gen-Ral coils, made by General Mfg. Co.

A-Shielded R.F. transformer having a bank-wound secondary made of litz wire, with an ordinary single-layer primary wound over the lower end of the secondary. With the average pentode tube this transformer gives a gain of 42 at 550 kc. and a gain of 62 at 1,500 kc.; this nearly uniform gain over the tuning range is secured by proper design of the secondary coil. You will find this coil used in circuits like the tuned secondary transformer load arrangement of Fig. 8A.

B-Shielded I.F. transformer. Triplesection cross-wound primary and secondary coils are used to give a high Q factor. The coupling (spacing between coils) is adjusted during manufacture for optimum results. Tuning condensers are built into the housing, one being connected across

and C_2). A stiff wire attached to the antenna post (the high R.F. terminal) of primary coil L_1 loops around the grid end of secondary coil L_2 . This arrangement gives practically uniform gain over the entire tuning range. You will find this stiff wire scheme used for



each coil. Look for units like this in double-tuned transformer load circuits such as that in Fig. 9A.

C-Shielded I.F. transformer. The crosswound coils are made with litz wire and are weakly coupled (coefficient of coupling is less than the critical value) to give sharp peak response. Also used in circuits like Fig. 9A.

D-Shielded R.F. transformer having cross-wound primary and secondary coils mounted permanently on a wood dowel; coupling cannot be adjusted. This construction givbs fair gain, but this varies considerably over the tuning range. Used in circuits like Fig. 8A.

E-Unshielded antenna coil for broadcast band. Secondary is made of litz wire, bank-wound directly on coil form, while the lattice or cross-wound primary is located over one end of the secondary. The primary is self-tuned (hy its distributed capacity) to about 550 kc., and a heavywire coupling ring provides capacity coupling between primary and secondary, so that the gain is very nearly uniform throughout the broadcast band. Fig. 13 illustrates how the coil is used.

capacity coupling between coils in practically all of the small universal A.C.-D.C. tuned radio frequency receivers. (In some receivers this stiff wire is replaced by several turns of insulated wire or by two short insulated leads twisted together.)

R.F. Tuning Circuits Used in Actual Receivers

Although the R.F. tuning circuit diagrams already presented in this lesson are typical of those used in most radio receivers, there are a number of minor variations of these circuits which will occasionally be encountered. Three different receivers which use out-of-the-ordinary R.F. tuning circuits have been selected for study. The circuit diagrams of the R.F. sections of these sets are shown in much the same way as they appear in the rely does not have A.V.C.; the volume is controlled by a dual rheostat which simultaneously varies the resistance across the antenna input coil (this is done by R_1) and varies the C bias voltage on the R.F. tubes (this is done by R_2). The cathode, screen grid and plate by-pass condensers for each tube are mounted in a single case (indicated by dotted lines and labeled C_P).

You will readily recognize L_1 - C_1 as the first tuning circuit, this being fed with the antenna signal by a direct connection from the antenna to a tap



FIG. 14. A portion of the circuit diagram of the General Motors model MA T.R.F. receiver, illustrating an interesting method used to equalize the gain over the entire tuning range.

spective service manuals, in order to make you familiar with the different drawing techniques used in actual practice.

General Motors Model MA T.R.F. Receiver. The first two stages of this receiver are shown in Fig. 14; in the actual receiver there are three R.F. stages, giving four tuned circuits which can be tuned over the 550 kc. to 1,500 kc. broadcast band by a single tuning control. Observe that type 24 screen grid tubes are used; since these have very high A.C. plate resistance values, their A.C. plate currents can be considered essentially constant under all circuit conditions. This receiver clearat point 1 on L_1 . But notice that the antenna signal must first pass through coil L_{A} , which is not inductively coupled to L_1 ; we suspect immediately that this is used by the designer to equalize the gain over the tuning range. Coil L_{A} actually serves to tune the antenna at the low frequency end of the tuning range, reducing the antenna impedance to a minimum value and thereby causing a large signal current to flow into the portion of coil L_1 between point 1 and ground. At the higher frequencies in the range, the antenna is off tune. antenna current is lower and the gain of the input tuning circuit is consequently lower also. Thus coil $L_{\mathbb{A}}$ can

counteract a tendency for tuning circuit L_1-C_1 (or any other tuning circuit in the receiver) to give higher gain at the higher frequencies.

Now let us analyze the second tuning circuit, made up of L_2 - C_2 . The voltage developed across R.F. choke L_P by the flow of A.C. plate current through this high-impedance choke is applied to this second tuning circuit through coupling condenser $C_{\mathbf{K}}$. Coil L_P has a certain amount of distributed capacity which tunes it to resonance at a low frequency in the tuning range, thereby raising its impedance at low frequencies and increasing the signal voltage which it can deliver to the second tuning circuit.

Since the reactance of coupling condenser C_{κ} decreases at the higher frequencies, this condenser will transfer more voltage at the higher frequencies. Coil $L_{\rm P}$ thus serves to raise the gain at low frequencies, while C_{κ} raises the gain at high frequencies; observe that C_{κ} is adjustable so that any amount of equalization of gain over the tuning range can be attained. This condenser need be adjusted only as a part of the regular alignment procedure for the receiver.

General Electric Model F-107 Superheterodyne Receiver. You know that the higher the coil Q factor in a tuning circuit, the higher is the gain and the selectivity of the circuit. Sometimes, however, the circuit designer runs into a condition where he has exactly the desired selectivity but has too much gain. To reduce this gain in order to prevent overloading, he could use shunt resistors across tuning condensers, but this would also reduce selectivity and would therefore be undesirable. One designer solved this problem by choosing R.F. coils which gave the desired selectivity, then tapping one of the coils to remove only a fraction of the total available voltage. The result is the I.F. amplifier section illustrated in Fig. 15, where the tap at point 1 allows only about 60% of the total available voltage in this tuning circuit to be impressed upon the grid of the second I.F. tube. If necessary, an even greater reduction in gain could be secured by using a similar tap for the plate connection on the primary winding of this I.F. transformer.

RCA-Victor Model 5M Auto Radio Receiver. An antenna input coupling circuit which is widely used in auto radios is shown in Fig. 16. The filter circuit made up of C_1 , L_1 and C_2 is known as a low-pass filter and is used



FIG. 15. Here is one J.F. stage of the model F-107 General Electric receiver, an example of a fixed frequency amplifier. This circuit diagram (taken from the service manual for the set) tells how the I.F. transformer leads in this receiver can be identified by their color. Note that the D.C. resistance of each coil is indicated, for continuity testing purposes.

to allow only R.F. signals which are below 1,500 kc. to pass. This is highly important in an auto radio receiver, for the spark coil type ignition systems used in automobiles develop ultra-high frequency radio waves which can create interfering signals if they enter the input circuit of the radio receiver. R.F. signals in the desired frequency range pass through this filter without appreciable opposition and develop a signal voltage across input coil L_2 . This signal voltage is in turn fed through coupling condenser C_3 into the first resonant circuit, the coupling being through that section of coil L_3 which is below point 1 and then through condenser C_{\bullet} to ground. As a result there

is both inductive and capacitive coupling, giving equalization of gain over the entire tuning range. Tuning condenser C_5 tunes the portion of L_3 between point 1 and the grid of the first tube over the entire tuning range; C_4 is simply a trimmer condenser used for alignment purposes. The three vertical lines and one diagonal line above coil L_3 symbolically indicate that the coil uses a pulverized iron core, with the diagonal line indicating that the position of the core and therefore the inductance are adjustable. This core thus provides an aligning adjustment at the low frequency end of the tuning

to equalize the gain; in a circuit of this type, we are safe in assuming that the primary will tune to a lower frequency in the range to counteract increasing gain provided by the secondary circuit at the higher frequencies. The low resistance of the secondary coil indicates that this coil has a high Q factor.

Conclusion. I do not expect you to be able to analyze in a few minutes any R.F. tuning circuit which you may encounter after your study of this lesson, for there apparently is no limit to the number of variations which are being used by circuit designers. I do say, however, that after you have studied a



FIG. 16. Input circuits of the model 5M RCA Victor Auto Radio. Condenser C1, labeled SPARK PLATE, provides a shunt path to ground for high-frequency interference signals produced by the spark plugs and ignition system of the automobile.

range, while high frequency trimmer C_4 permits alignment at the high frefrequency end of the range.

Turning now to the plate circuit of the first tube, you might say at first glance that this contained a conventional tuned secondary transformer load. Closer examination would show, however, that while the D.C. resistance of the primary coil was 80 ohms, the secondary coil had a D.C. resistance of only 7 ohms. These figures tell us that the primary coil must be made of many turns of fine wire and hence must have a high inductance and high distributed capacity. This distributed capacity tunes the primary to some frequency in the tuning range in order

number of other schematic circuit diagrams in the same way as we have analyzed the three diagrams at the end of this lesson and after you have acquired practical experience with the behavior of tuning circuits in actual radio receivers, you will find yourself in a far better position to service radio receivers and know what to expect from them than if you simply made adjustments blindly without knowing exactly what you were doing. You will know why certain things happen when certain screws and knobs are turned in a receiver, and will be able to determine whether or not the fidelity or selectivity of a particular receiver can be improved.

Lesson Questions

Be sure to number your Answer Sheet 22FR-2.

Place your Student Number on every Answer Sheet.

Never hold up one set of lesson answers until you have another ready to send in. Send each lesson in by itself before you start on the next lesson. In this way we will be able to work together much more closely, you'll get more out of your Course, and you will receive the best possible lesson service.

- 1. What does a sharp peak response curve for an R.F. amplifier indicate as regards gain and selectivity?
- 2. Will an R.F. amplifier which has a sharp peak response curve cause severe attenuation of the higher modulation frequencies?
- 3. Will amplitude distortion occur when an R.F. amplifier has a symmetrical double-peak response curve?
- 4. How is a high Q factor obtained for a coil which has a given inductance?
- 5. Does the Q factor of a practical coil vary with frequency?
- 6. Why can a pentode tube in the R.F. amplifier stage of a receiver be considered as a constant-current generator?
- 7. What effect does the loading of a tuned circuit in an R.F. amplifier have upon its selectivity and gain?
- 8. Name two advantages which are secured by using a double-tuned transformer load circuit.
- 9. Is it possible to adjust the trimmer condensers for a single-peak response when the two coils in a double-tuned circuit are over-coupled?
- 10. If the peaks in the response curve of a double-tuned circuit are excessively high with respect to the valley between them, how can they be reduced without changing the band width?

THE VALUE OF REVIEW

Man has acquired so much new knowledge in recent years that it has become impossible for one person to know even a small fraction of the available information. Educational authorities realize this fact, and the colleges of today consider a man welleducated if he knows the elementary ideas and knows where to find other information when he wants it.

Radio, along with the other fields of endeavor, has outgrown the memorizing ability of the human mind. Also, radio is such a comprehensive field that occasionally you cannot recall important facts previously studied. Review is obviously the solution to this problem.

Time spent in review several weeks or months after a book is studied will be far more profitable than an equivalent amount of extra time spent on the book initially, for your mind has then had a chance to file and store away the information secured from the first study. Each review results in more information being transferred from the textbook to your mind, and soon, with no conscious attempt to memorize, you will find yourself able to recall an amazing number of valuable facts.

J. E. SMITH



23FR-3

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STUDY SCHEDULE NO. 23

This textbook is crammed with practical facts, practical circuits and practical servicing hints for superheterodyne receivers, most of these being based upon radio theories presented in previous lessons. This means that, if any subject in this lesson gives a bit of difficulty, you can clear up things by reviewing the previous lesson which covers the theory of that subject. Thus, if the operation of the local oscillator in a superheterodyne is not quite clear to you, review the lesson on oscillators.

- - 3. The Local Oscillator Pages 14-17 There are a number of requirements placed on the local oscillator. It has to supply a voltage at the proper frequency, without frequency drift, and without fluctuations in amount. Typical circuits are shown.
- - - 8. Mail Your Answers for this Lesson to N.R.I. for Grading.
 - 9. Start Studying the Next Lesson.

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HOW BROADCAST, ALL-WAVE, AND TELEVISION SUPERHETERODYNE RECEIVERS WORK

The Superiority of the Superheterodyne

THE superheterodyne receiver is by far the most common type of radio set in use today, so it is well worth further study. You have learned, in previous lessons, how almost all the components of a "superhet" function and have been given a preview of this circuit. Now we will collect this information and show you in more detail how the various components act when they are combined. If you find you are a little hazy on the operation of some individual circuit component, by all means make a quick review of the lesson in which it was discussed. Doing so will make it much easier for you to grasp all the details of the operation of the superheterodyne.

Basically, the superheterodyne principle of r.f. amplification involves the conversion of each incoming signal within the receiver tuning range to one definite fixed radio frequency which is known as the intermediate frequency or the i.f. value.* It is much easier to get optimum (best) results from an r.f. amplifier which always works at the same frequency, as in the i.f. amplifier in a superheterodyne receiver, than from an amplifier which is tuned through a wide range of radio frequencies, as in t.r.f. receivers. ► This is true for frequency modulation receivers as well as for amplitude modulation receivers. Incidentally, both use the same sections—the important differences are that an FM receiver has a slightly different i.f. design, uses a limiter stage, and has a different demodulator or second detector. (FM receivers are described fully in



This large console radio uses the superheterodyne circuit.

another lesson.) This lesson covers AM receivers; however, you will need all this information to gain a complete understanding of FM receivers.

Review of R.F. Amplifier Operation. If you will review in your mind the action of an r.f. amplifier, you will recall that the gain and selectivity of

^{*}When a signal demodulator (second detector), audio amplifier, loudspeaker, and power pack are added to a superheterodyne r.f.-i.f. amplifier, the result is a complete superheterodyne receiver. The above-mentioned sections of the receiver are of conventional design and are all covered thoroughly elsewhere in the Course.

any stage are dependent upon the resonant resistance of the plate load; this resonant resistance depends upon L/C (the ratio of coil inductance to condenser capacity), and upon the losses in the tank circuit. In tuning to different frequencies, we vary the condenser capacity and thus vary the L/C ratio. For this reason, and because tank circuit losses are different at each frequency, uniform amplification at all frequencies is not obtained readily. Thus the amplification and the selectivity of a t.r.f. receiver are different at each setting of the tuning dial.

▶ In the superheterodyne receiver, on the other hand, practically all of the gain and selectivity are produced at one fixed i.f. value, regardless of the frequency of the incoming signal; selectivity and sensitivity are therefore uniform at all tuning dial settings.

A comparison of the performance curves of a superheterodyne receiver with those of a t.r.f. receiver having approximately the same number of tubes will show very clearly the superiority of the superheterodyne. Such curves are given at A, B, C and D in Fig. 1. In these, sensitivity is measured in terms of microvolts of signal input required to get a 50-milliwatt output; thus, a lower input value means a more sensitive receiver. The sensitivity eurve for the superheterodyne (Fig. 1A) is practically constant for all frequencies in the tuning band, while the same curve for the t.r.f. set (Fig. 1B) varies



FIG. I. These sensitivity and selectivity curves for a superhaterodyne receiver and for a t.r.f. receiver show clearly the superiority of the superheterodyne.

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considerably and represents much lower sensitivity. The selectivity curves in Figs. 1C and 1D give, for various frequency values off resonance, the ratio of signal input at resonance to the signal input required to give the same output at the off-resonance frequency: the steeper the curve, the more selective the receiver. The curve in Fig. 1C (for the super) thus represents very good selectivity which is uniform at all frequencies in the band, while the curves in 1D (for the t.r.f. receiver) show poor selectivity at 600 kc., and show increasingly poorer selectivity at higher frequencies in the band.

 \blacktriangleright If a t.r.f. receiver were designed for all-wave reception, it would be necessary to change every coil in the set each time the band* was changed. In a superheterodyne receiver, however, only the coils in the preselector and oscillator sections need be changed. Because of this, every practical all-wave receiver is a superheterodyne. In addition, the public demand for high-fidelity reception, which entails amplification of wide side bands over the entire range of carrier frequencies, is easier to meet with the fixed i.f. amplifiers used in supers. It is no wonder that the superheterodyne is the most widely used system today; only a few of the compact and inexpensive midget receivers are manufactured now with t.r.f. circuits.

PREVIEW OF SUPERHETERODYNE PRINCIPLES

Before the i.f. amplifier can do its job, the incoming signal with its picture or sound modulation must be converted to the intermediate frequency selected. This is done by combining the *incoming modulated r.f. carrier* with a local r.f. oscillator signal—a process called "frequency conversion."

Now, when two signals of different frequencies are mixed together and sent through a detector or demodulator tube circuit, the plate current will consist of many components. There will be signals at the two original frequencies, a signal whose frequency is the



Courtesy Emerson Radio and Television Corp.

Battery-operated portables like this use the same basic superheterodyne circuit as is used in powerline operated sets, except for the use of battery type tubes.

difference between the original frequencies, a signal whose frequency is the sum of the original frequencies, and harmonics of all four of these frequencies. Each of these plate current compopents will carry the original modulation. By placing a highly selective resonant or tank circuit in the plate circuit of the detector, we can tune to any signal component and thus separate it from the others. Inasmuch as it is easier to make selective high-gain amplifiers for low r.f. values, the difference between the two original frequencies is always selected as the i.f. value

If, for example, the incoming frequency is 1000 kc. and the local r.f. os-

^{*}The range of frequencies which can be covered by a given combination of a coil and a condenser when one of them is variable is called a *band*.

cillator signal frequency is 1175 kc., the plate current of the detector circuit will contain these two frequencies, as well as 2175 kc. (the sum frequency) and 175 kc. (the difference frequency). From what we just said, the 175-kc. frequency will be selected as the i.f. value. The following stage (the i.f. amplifier) will be designed to pass and amplify only this 175-kc. signal and such side frequencies as are required to give the desired fidelity characteristic to the receiver.

Now, when a 500-kc. signal is tuned in, a local oscillator signal of either 325 kc. or 675 kc. will produce a difference frequency of 175 kc. If the incomit is difficult to tune a coil and condenser combination over a range having a frequency change which is greater than about 3.3 to 1. For this reason, the local oscillator in a superheterodyne is generally made to produce a frequency higher than that of the incoming signal. In our example, then, the oscillator would vary in frequency from 675 kc. to 1675 kc. as the receiver was tuned from 500 kc. to 1500 kc.

► Turning now to a block diagram of a superheterodyne type r.f. amplifier (Fig. 2), we find that there are four important sections: 1, the preselector; 2, the local oscillator; 3, the mixer-first detector; and 4, the i.f. amplifier. Be-



FIG. 2. A block diagram showing the four important r.f. sections of a superheterodyne; the second detector and the stages following it are identical for both t.r.f. and superheterodyne receivers, and hence are not discussed in this book.

ing signal is 1500 kc., a local oscillator signal of either 1675 kc. or 1325 kc. will produce the required i.f. value. This brings up the question of whether we should make the oscillator frequency lower or higher than the signal frequency.

▶ Let us consider this problem for a receiver which is to tune from 500 kc. to 1500 kc., with the i.f. at the fixed value of 175 kc. Clearly, the local oscillator must vary in frequency either from 325 kc. to 1325 kc. or from 675 kc. to 1675 kc. In the first case (325 to 1325) there is a 4 to 1 change from the highest to the lowest frequency, and in the second case (675 to 1675) there is a 2.5 to 1 change. As a practical matter,

fore we study these sections in detail, let us see what the important functions of each are.

Preselector. The preselector consists of one or more resonant circuits which can be adjusted to the frequency of the desired r.f. signal; it may or may not have r.f. amplifier stages. Theoretically, the preselector is not necessary in order to convert an r.f. signal to a lower frequency value. If you omit it, however, interfering signals will get into the mixer-first detector and they will react with each other or with the local oscillator signal to produce undesired signals at the i.f. value. Some gain (amplification) in the preselector is always desirable, because the mixer-

first detector creates inherent noise which can be over-ridden only by a strong r.f. signal.

Local Oscillator. Frequency conversion cannot take place unless a local signal is produced, which differs from the incoming r.f. value by the i.f. value. A local oscillator is therefore of fundamental importance in the superheterodyne circuit. The preselector tuning condenser and the oscillator tuning condenser either can be controlled independently or can be ganged together and controlled by a common tuning dial. The latter practice is now universally followed, for with it the oscillator is always at the correct setting to deliver a signal differing from the incoming signal by the i.f. value. Besides giving single-dial tuning, ganging together the oscillator and preselector tuning condensers reduces what is known as repeat point reception—a phenomenon which we will discuss a little further in this lesson.

Mixer-First Detector. This is the actual point of frequency conversion. Of great importance is the ability of this section to act as a detector for both the incoming and local oscillator signals, provided they are of reasonable intensity. Its plate circuit must contain a highly selective, high Q tank circuit having a resonant frequency equal to the i.f. value (the difference frequency), so this frequency can be accepted and the other plate circuit components rejected. This tank circuit should have a low L/C ratio, so the resulting high capacity value will then act as a by-pass for the unwanted signals.

I.F. Amplifier. Here the i.f. signal, modulated with the original picture or sound signal, gets its real boost in gain. The i.f. amplifier must be able to amplify the i.f. signal and some or all of the important side-band frequencies, depending upon the type of receiver performance desired. The i.f. amplifier can be made highly selective, thus cutting out undesirable signals as well as unimportant side-band frequencies of the radio signal being received, or it can be made to have broad band-pass characteristics, thus passing a wide range of side frequencies. In the first case the i.f. amplifier is said to have good adjacent channel selectivity, and in the second case it has high-fidelity response characteristics.

Now that you've had a "preview" of the action of a superheterodyne r.f. amplifier, let's take up each section in detail. We will start with the preselector.

The Importance of the Preselector

It is possible to impress a desired signal directly (without tuning) on the grid-cathode terminals of the mixerfirst detector tube in a superheterodyne r.f. amplifier circuit and to secure, with the aid of the local oscillator, a beat frequency output signal having the desired i.f. value. Of course this is never done in practice, because there are a number of undesirable results. By considering first the effects encountered in a theoretical direct-input circuit (one having no preselector), we can learn a great deal about the importance of the preselector and about the problems which may be met in service work.

Direct Input Circuit. The mixerfirst detector section of a superheterodyne circuit which has no preselector is given in Fig. 3; any signal f_s which is picked up by the antenna flows through resistor R to ground, and the r.f. voltage developed across this resistor is fed *directly* to the grid of the tube. The local oscillator feeds into the cathode circuit of the tube a signal which we will designate as f_0 . Assume that the frequency of this signal can be independently controlled by varying the setting of oscillator tank condenser C_{o} . The resonant circuit in the plate lead of the tube is adjusted to the desired i.f. value $f_{1.F.}$, so that only the i.f. current produces a voltage drop across the i.f. resonant circuit for further amplification. For the present, we need not consider any other parts or sections of this superheterodyne circuit.

REPEAT POINTS (DOUBLE-SPOT TUNING)

Assume that only one signal, having a frequency of 1000 kc., is being picked up by the antenna in Fig. 3, and that the i.f. resonant circuit is adjusted to an i.f. value of 100 kc. Under these conditions the required 100-kc. beat frequency $f_{I.F.}$ will be produced when the oscillator is tuned to 1100 kc. But we can also secure this 100-kc. beat frequency by setting the oscillator to 900 kc. Thus, there are two oscillator tuning dial settings at which the 1000kc. incoming signal will be passed on to the i.f. amplifier. This condition is called repeating or double-spot tuning, for we have repeat points-two different settings at which the same station can be received—on the oscillator tuning dial. These repeat points are present in any superheterodyne circuit when the oscillator can be separately tuned, even if resistor R is replaced with a highly selective preselector circuit. The repeat point for any one station is always separated from the correct oscillator dial setting by twice the i.f. value.

► Of course, superheterodynes with separately tuned oscillators have long been obsolete. Ganging the preselector and oscillator will give single dial control and, in addition, will serve to eliminate repeat-point reception provided a good preselector is used.

Service Hints. A repeat point can occur in a single-dial receiver if the selectivity of the preselector is too low or is impaired by a circuit defect or improper adjustment, or if the signal from the station is exceedingly strong. In each case, you would hear a station at its correct dial setting and at a repeatpoint setting which will be *below* the correct dial frequency setting by twice the i.f. value (assuming that the oscillator tunes *above* the frequency of the incoming signal). Examine the set for defects and check the alignment. If there appear to be no defects and the alignment is normal, then the condition may be due to the design of the receiver. If no interference occurs with another desired signal, then you can ignore the condition, as it is impractical to change the receiver design. (Explain it to the customer if he remarks about it.) However, if there is interference, you should follow the hints in the next Section of this lesson.

IMAGE INTERFERENCE

From your own experience you know that signals of many different frequenthe desired signal by twice the *if*. value, is heard along with the desired signal. This condition is called *image* interference.

Notice that image interference is caused by repeat-point reception of an *undesired* station—in fact, a repeatpoint signal could be called an image. The only difference is that now the undesired signal is interfering with a desired signal.

► The obvious solution to this problem is to use a highly selective preselector which is capable of tuning to the desired signal and of rejecting the



FIG. 3. Schematic diagram of the mixer-first detector section of an imaginary superheterodyne receiver which has no preselector.

cies are always present in the antenna circuit of a receiver. Let us, therefore, assume that in addition to the desired 1000-kc. signal, there is an undesired 1200-kc. signal in the antenna circuit of Fig. 3. The oscillator is set at 1100 kc. in order to convert the desired signal to the i.f. value of 100 kc., but the undesired 1200-kc. signal can also mix with the 1100-kc. oscillator signal and produce a 100-kc. beat frequency. Both the desired 1000-kc. signal and the undesired 1200-kc. signal will then get through the i.f. section, be reproduced by the loudspeaker, and cause interference. This occurs when an interfering signal, whose frequency is above image signal. However, the ideal preselector, which will allow only a single frequency or a narrow band of frequencies to pass and will absolutely reject all other frequencies, does not exist. An engineer is quite satisfied if he can design a preselector which *reduces* the strength of the interfering station (at the image frequency) 1000 times.* This number is called the *image interference ratio* and, in this case, means

^{*}This ratio value must be considerably higher in receivers which have high-gain if. amplifier sections. In all-wave receivers, however, ratios as low as 100 to 1 for the higher frequency bands are considered acceptable because it is difficult to design all-wave sets with better ratios.

that an undesired image signal (of a strength equal to that of the desired signal) will be heard 1000 times weaker than the desired signal.

► Naturally, the closer the desired and undesired frequencies are to each other, the greater the problem is of separation. When a low i.f. value is used, it often takes two or even three preselector tuned circuits in cascade to get an image interference ratio of 1000, and



A typical table-model superheterodyne receiver.

this increases the cost of building the receiver. Special image-rejecting circuits have been developed, but a simpler solution involves the use of a high i.f. value, somewhere between 250 and 500 kc.

Let us see what advantage is secured by using a high i.f. value. Assume that a certain receiver which is tuned to a 1000-kc. signal has an i.f. of 500 kc. The oscillator will therefore be at 1500 kc., and the image interference frequency will be 2000 kc. Only a simple preselector tuned to 1000 kc. is needed here to reject the undesired 2000-kc. signal, since the difference between the frequencies of desired and undesired signals is so great. On the other hand, if the i.f. value is 150 kc., the oscillator will be at 1150 kc., and the image will be at 1300 kc. This is much closer to 1000 kc., and will require a far better

preselector. Thus, superheterodynes having high i.f. values will have simple preselectors, while highly selective preselectors are imperative for sets with low i.f. values (values between about 185 kc. and 250 kc.).

Local stations of image frequency are sometimes so powerful, however, that even an image interference ratio of 1000 is insufficient to prevent interference with some desired station.

Service Hints. When a receiver is to be serviced for image interference, the first step would be to realign the preselector. Shortening the antenna is another remedy, for a very long antenna broadens the response characteristic of the first resonant circuit in the preselector, thus letting the undesired signal get through. If only one station is causing image interference, place in the antenna circuit a wave trap which is tuned to the frequency of this station.

Another possible solution would be to shift the i.f. value about 10 kc. This would move the repeat point to another frequency, away from the desired frequency. In this case, the repeatpoint still exists but as long as interference with a *desired* signal is eliminated, it will not matter greatly. However, it is not always possible to make this change. It will interfere with the tracking when the oscillator tuning condenser has specially-cut plates, but it can be carried out if padder condensers are used. (You will study oscillator tracking later in this lesson.)

INTERMODULATION INTERFERENCE

When any two signals whose frequencies differ by exactly the i.f. value exist in the antenna circuit of Fig. 3, they can beat with each other, producing the i.f. value without the aid of the local oscillator. This condition is known as intermodulation interference. Without a preselector, these mixed signals cause an interference sounding like garbled (unintelligible) speech which would be heard regardless of the oscillator dial setting. When a preselector is used, only those r.f. signals which get through the preselector can produce this trouble, and this reduces the number of possible interfering signals greatly.

Service Hints. If garbled speech continues even when the local oscillator is blocked or is detuned, intermodulation interference is present. Improving the selectivity of the preselector blocks out the interfering stations and thus eliminates the trouble. If this is impractical, try cutting out one of the interfering stations with a wave trap; try a shorter aerial; and try changing the i.f. value of the receiver about 10 kc. (or to a value ending in 5).

OSCILLATOR HARMONIC

We have assumed up to this time that the local oscillator is feeding only its fundamental frequency to the mixer-first detector. In many cases, especially where the oscillator and its coupling circuit are poorly designed, harmonics of the oscillator (usually only the second harmonic) may reach the mixer-first detector and react with an undesired incoming signal to produce an undesired i.f. beat signal. This condition is called oscillator harmonic interference. For each oscillator setting there may be two frequencies which incoming signals can have in order to beat with the second harmonic of the oscillator and produce an undesired i.f. signal. An example will explain how this occurs.

Suppose that a receiver which has a 260-kc. i.f. value is tuned to an 1160-kc. station. The oscillator fundamental frequency will be 1160 + 260, or 1420

kc. and the second harmonic of this will be 2840 kc. Now any signal, which differs from 2840 kc. by 260 kc. and which is strong enough to get through the preselector, will produce an i.f. beat frequency which can cause interference. Thus either an aircraft radio station at a frequency of 3100 kc. (2840 + 260) or a commercial station on 2580 kc: (2840 - 260), or both, could be heard on this set with the desired 1160-kc station.

You can identify oscillator harmonic interference if you can identify the interfering station as being one with a frequency either above or below the second harmonic of the oscillator by the i.f. value. (Oscillator harmonics higher than the second are so weak that they can be neglected.) The frequency difference between interfering and desired signals is so great in the case of harmonic interference that generally only strong local stations, such as amateur, police, commercial, or government code or phone stations, can ride through the preselector.

Service Hints. Either improving the selectivity of the preselector by realigning it to keep out the interfering signals, or adjusting the voltages of the oscillator to suppress its second harmonic are possible remedies for harmonic interference. However, the installation of a wave trap which is tuned to the frequency of the offending station is the simplest cure. When the interfering signal is especially strong, it may be necessary to shield the mixerfirst detector to prevent the signal from acting on it directly without going through the preselector and wave trap: a filter in the power line may also be needed.

CODE INTERFERENCE

Trouble can be caused also by an undesired signal having a frequency equal

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to the i.f. value of the receiver. If this signal gets through the preselector, it will be passed on by the mixer-first detector to the i.f. amplifier, as there is no need for frequency conversion. Since most transmitters below 500 kc. (in the range of i.f. values) are code stations, this trouble is commonly referred to as code interference. It may be heard at any point on the tuning dial, but is strongest at the lower end of the receiver dial, around 550 kc.

Occasionally the second harmonic of a powerful local long-wave transmitter tuning dial can be eliminated by installing a wave trap which is tuned to the interfering code station, by shortening the antenna, or by changing the i.f. value of the receiver.

▶ Bear in mind that the cures suggested here and in the previous service hints are necessary because of the inability of the preselector in the receiver to keep out undesired signals. These cures were suggested only because the fundamental cure, *redesigning* the preselector for greater selectivity, is usually impractical.



FIG. 4. A simple tuned r.f. transformer preselector is shown at A. A wave trap is added in B.

may produce code interference. For example, if a receiver has an i.f. of 480 kc., a local 240-kc. station might be heard at all points on the tuning dial. This could be produced because the second harmonic of the 240-kc. carrier is getting through the preselector, but a more likely condition is that the 240kc. signal gets through the preselector and harmonics of it are created by the rectifying action of the mixer-first detector tube. The second harmonic, being of the i.f. value and carrying the modulation of the original code signal, would pass through the i.f. amplifier and produce interference.

Service Hints. Code interference which is heard at all settings of the ► Having studied the importance of the preselector, let us take up a few of the common preselector circuits.

R.F. TRANSFORMER WITH A TUNED SECONDARY

The simplest of all preselectors is the tuned-secondary r.f. transformer shown in Fig. 4A. It is often quite satisfactory in a receiver which uses a high i.f. value. The selectivity of this circuit is essentially dependent upon the mutual inductance M and upon the frequency of the desired incoming signal; increasing either reduces the selectivity. In a receiver which is to be used with a short antenna, the mutual inductance usually is made quite large so that a strong input signal can be obtained, but such a receiver tunes broadly (has poor selectivity) when coupled to a long antenna. Even with a high i.f. value, such a circuit will tune broadly at very high frequencies (around 20 megacycles). It is for this reason that all-wave receivers ordinarily require better preselector circuits than this.

Wave-Trap Circuit. A wave trap is often connected in series with the primary of the r.f. transformer, as shown in Fig. 4B, to eliminate the interference caused by a particular station. C_{T} - L_{T} is the wave trap; it is tuned to the frequency of the interfering station by adjusting trimmer condenser C_{T} . Factory-installed traps usually tune to the i.f. value, to eliminate code interference, which is one of the most annoving and prevalent of the interferences. (A serviceman can connect a similar trap in the antenna lead of any radio receiver having image interference, harmonic interference, intermodulation, or code interference troubles caused by one station. The trap is tuned to the frequency of the interfering station.)

BAND-PASS PRESELECTORS

In the preselector circuit just considered, only one tuned circuit contributes to the selectivity of the receiver. An extra tuned circuit connected in cascade as shown in Figs. 5A, 5B, and 5C will greatly improve the selectivity and, consequently, will reduce interference troubles. This extra circuit also permits adjustments which give band-pass characteristics to the preselector. The circuit shown in Fig. 5A is simply the circuit of Fig. 4A with an additional resonant circuit, made up of L_{s2} and C_2 , coupled by mutual induction to L_{s1} . Fig. 5B shows two resonant circuits which are directly coupled inductively, with coil section L_{M2} common to both resonant circuits. Capacitive coupling is used in the circuit of Fig. 5C, with condenser $C_{\rm K}$ common to both resonant circuits. For a fixed value of coupling, a single peak resonance characteristic is obtained when condensers C_1 and C_2 are tuned for maximum receiver output. Increas-





ing the coupling between the resonant circuits or tuning C₁ above and C₂ below the resonant frequency gives a flat or double-peak response characteristic. When this is done, the circuits are referred to as band-pass preselectors. ____ ► Although band-pass preselectors are quite effective in eliminating (or at least reducing) the many types of interference troubles, they have one important drawback in that they considerably reduce the strength of the incoming signal. One way of overcoming this loss in signal strength is to step up the gain of the intermediate frequency

resonant circuits which have a high Q factor, which operate at a high temperature, and which have a wide response characteristic. A hissing or frying noise which is heard in the loudspeaker and is especially loud when the receiver is tuned between stations is an indication that thermal agitation is present. Because of this effect, the practical limit to the sensitivity of a receiver is about 1 microvolt (which means that the smallest signal voltage which can be made to give 50 milliwatts output to the loudspeaker is approximately 1 microvolt).



FIG. 6. A widely used preselector circuit, in which one stage of r.f. amplification boosts the strength of the incoming signal before it reaches the mixer-first detector. This additional amplification makes the desired incoming signal over-ride any noise which may be present in the mixer-first detector, and also lessens interference troubles.

amplifier section, but two undesirable effects, thermal agitation and converter noise, become annoying when this is done. Let us consider them:

1. Thermal Agitation. Free electrons are continually moving around in any conductor, producing tiny pulses of electron current. Some component of these pulses will be at the same frequency as that to which the receiver is tuned. Hence, these pulses of electron current undergo resonance step-up, along with the signal currents. The effect of these pulses of electron current is commonly designated as *thermal agitation* of electrons. It is greatest for

2. Converter Noise. Even more troublesome than thermal agitation is an effect which occurs in the mixer-first detector tube. Although the average plate current value is controlled accurately by signal voltages, nonetheless, there are current variations caused by irregularities in the electron emission. As a result, electrons emitted by the cathode arrive at the plate in spurts or "shots." These variations are amplified by succeeding stages along with the desired signals, and are heard in the loudspeaker as a "frying" noise. This action is known as the shot effect, as the electron grain effect, or as frequency

converter* noise. Although the shot effect is present in practically all vacuum tubes, the variation in plate current due to it is ordinarily so small in comparison to the average plate current that the effect is negligible. In a tube which operates as a detector, however, the average plate current is so low, because of the high negative C bias, that the variations in current affect an appreciable part of the total plate current. Frequency converter noise is most noticeable when a receiver is tuned to a weak signal; strong signals tend to "drown out" or over-ride the noise. The strength of the signal fed to the input of the mixer-first detector must be large enough to make the signal-tonoise ratio at the output of this section as great as possible and thus minimize the effects of frequency converter noise. ► It is highly desirable to have a stage or two of r.f. amplification ahead of the frequency converter section in_ order to build up the strength of the incoming signal so that it will override any converter noise which is pres--ent in the mixer-first detector tube. The greater the signal strength with respect to the noise, the less disturbing will be the noise.

R.F. Amplified Type of Preselector. A widely used preselector circuit which contains a stage of r.f. amplification to increase the signal strength at the input to the mixer-first detector is shown in Fig. 6. The amplification of this stage must be made high enough to eliminate frequency converter noise. yet not so high that it increases the total gain of the receiver to the point where thermal agitation effects in the r.f. antenna transformer will come through. The first resonant circuit. consisting of L_1 and C_1 is sometimes replaced by a band-pass resonant circuit of the form shown in Figs. 5A, 5B, and δC , in order to increase further the image-interference ratio.

This amplifier stage is not intended to increase the gain greatly—the i.f. amplifier still is being depended upon for most of the amplification. However, even a small increase is helpful in increasing the sensitivity and in overcoming the frequency-converter noise. An r.f. stage like this is frequently found in all-wave receivers.

Another advantage of the r.f. stage is the extra resonant circuit it provides in the preselector. (Both the r.f. and the mixer stages have resonant input circuits.) This extra selectivity helps to improve the image rejection ratio, thus reducing the interference.

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^{*}The mixer-first detector and the local oscillator together constitute the frequency converter section.

The Local Oscillator

If the local oscillator in a superheterodyne receiver is to perform its job satisfactorily, it must meet the following requirements:

- 1. At any dial setting, the frequency of the oscillator must be constant in value (there must be very little *frequency drift*).
- The voltage which the local oscillator supplies to the mixer-first-detector must be at least ten times greater than the voltage of the most powerful signal which is fed to the mixer-first detector by the preselector.
- 3. The variation in oscillator voltage output must be as small as possible as the frequency of the oscillator is changed by tuning (since absolute constancy is practically impossible, a maximum variation of 3 to 1 is considered by engineers to be satisfactory).
- 4. The output of the oscillator must have negligible harmonic content.
- 5. The oscillator itself should not radiate radio waves which would interfere with nearby receivers.
- 6. The oscillator must be coupled to the mixer-first detector in such a way that the frequency of the oscillator is not affected by changes in other receiver circuits.

The reasons for some of these practical requirements are important enough to warrant further explanation, so let us discuss these points before taking up typical circuits.

FREQUENCY DRIFT

The frequencies of transmitting stations are kept within extremely close

tolerances by special control devices: Government laws require that the frequency of a broadcast station shall not vary to any appreciable extent. Therefore, the signal which the preselector handles is quite constant in frequency. However, any variations in the local oscillator frequency while the receiver is tuned to a station will cause the beat frequency output of the first detector to vary in frequency. If this should occur and if the i.f. amplifier is sharply tuned, the i.f. amplifier will cut off varying amounts of side frequencies, which will result in distortion and weakened receiver output. This condition can be corrected by retuning. this changes the oscillator fre-88 quency. However, if the drift is large, you will soon reach a point where retuning has so changed the preselector resonant frequency that it cuts sidebands. Hence, this correction will work only over a limited range.

► When the oscillator operates at very high frequencies, as in an all-wave superheterodyne receiver, very slight circuit changes produce large amounts of oscillator frequency drift. This causes much trouble unless the response of the i.f. amplifier is broadened. Since this broadening will cause a loss in the adjacent channel selectivity of the receiver, it is better to use an oscillator which does not vary appreciably in frequency at any setting. Good frequency stability is secured by using in the oscillator a tank circuit which has a high Q factor and by limiting the loading of the tank circuit. Other frequency stability requirements include constant d.c. supply voltages, the locating of all oscillator parts away from any sources of heat, and the mounting of parts in such a way that

they will not be set into vibration by the loudspeaker.

OSCILLATOR OUTPUT VOLTAGE VALUES

To understand why the voltage which the local oscillator feeds into the mixer-first detector must be at least ten times greater than the signal input voltage to the mixer-first detector, we must consider the action of the two types of first detectors (square law and linear) which are commonly used in superheterodynes.

1. Square Law Type of First Detector. The incoming signal in a superheterodyne, as you know, consists of an r.f. carrier frequency and many r.f. side frequencies. Each of these must beat with the local oscillator to produce, after detection, the desired i.f. carrier frequency and its side frequencies. When a square law type of first detector is used, the strength of the beat frequency depends upon the product of the strengths of the local and incoming signals. (This statement is the result of a mathematical analysis of the problem.) The greater the voltage supplied by the local oscillator. then, the stronger will be the desired beat frequencies. However, the oscillator output voltage must not be made so large that the detector operates outside the square law region of its curve. This is why a ratio of about 10 to 1 between the voltages which the oscillator and the preselector feed into the mixer-first detector is highly desirable.

2. Linear Type of Detector. When a linear first detector is used, mathematics and experimental tests show that maximum i.f. output is obtained when the two signals (incoming and local) which are fed to the mixer-first detector are equal in strength. On the other hand, many harmonics of the beat (i.f.) frequency are produced under this condition. These harmonics, being two, three, four, etc. times the resonant frequency of the first detector plate load circuit will be tuned out here. But, they still may feed back into the input of the mixer-first detector through the plate-to-grid capacity or through stray coupling. This feedback energy will produce annoying audio beat notes (squeals) when the receiver is tuned to a station which has approximately the same frequency as one of these harmonics.

To eliminate this problem, it was found that if either one of the signals which are fed into the mixer-first detector is many times stronger than the other, the harmonic frequencies associated with linear detection become negligible. A ratio of 10 to 1 has been found sufficient in actual practice. Since it is easier to control the local oscillator than the carrier signal input strength, it is customary to make the signal voltage which the local oscillator feeds into the mixer-first detector at least ten times greater than the signal voltage fed into this section by the preselector regardless of the type of first detector used.*

Summary. With both types of detectors, then, satisfactory performance is obtained when the ratio of the local oscillator output to preselector voltage is greater than 10 to 1. When this ratio is less than 10 to 1, there will be a lowering of receiver gain (but no other undesirable effects) in the case of a square law detector. Also there will be

^{*}Since the incoming signal will vary considerably in strength if an r.f. amplifier stage is used in the preselector, it is important that when the first detector is of the linear type, an automatic volume control (a.v.c.) circuit is used to reduce the gain of the r.f. amplifier on strong signals and thus keep the ratio of the mixer input voltages higher than 10 to 1 at all times. A.V.C. circuits are taken up elsewhere in the Course.

an increase in receiver gain, accompanied by annoying squeals when a linear detector is used.

Variations in Oscillator Voltage. Once the amount of oscillator output has been chosen, it is desirable that this same amount be delivered, as near as possible, over the tuning range and in spite of operating voltage changes. Should the oscillator output increase



FIG. 7. Typical oscillator circuits.

greatly, the combined effect of local and incoming signals may be sufficient to swing the grid of the first detector positive. This would cause the grid to draw current and load the mixer input circuit, cutting down the peaks of the incoming modulated signal and thus producing distortion. About the only way we can have the local oscillator signal at least ten times as strong as the incoming signal at the mixer, and still keep the sum of these signals low enough so the grid cannot swing positive, is to be sure that fluctuations in oscillator voltage do not exceed a ratio of about 3 to 1. This means that the ratio of maximum to minimum oscillator output should not be more than 3.

Harmonics. The importance of preventing the local oscillator from feeding harmonics of its fundamental frequency to the mixer-first detector was considered earlier in this lesson, in connection with the study of harmonic interference.

Radiation. An oscillator needs only an antenna of some sort to become a midget radio transmitter. Obviously, it is the job of the radio engineer to see that such an antenna is not provided in a receiver. He does this by proper layout of circuit leads, and by using bypass condensers to prevent oscillator currents from leaking into any open wires which might serve as antennas and, in some cases, by shielding the oscillator coil.

TYPICAL OSCILLATOR CIRCUITS

Almost any oscillator circuit which has a fair degree of frequency stability will provide the local r.f. signal required in a superheterodyne receiver, but for practical reasons those circuits are used which require only one variable condenser, and which permit grounding of the rotor of the variable condenser. It is also desirable to have a circuit in which there is no high d.c. voltage across the variable condenser. Tuned grid and Hartley circuits are by far the more common, both in receivers having separate oscillator tubes and in those having combination oscillatorfirst detector stages. As a brief review, here are typical oscillators.

Tuned Grid Oscillators. Typical tuned grid oscillator circuits are shown in Figs. 7A and 7B, and in each, the frequency of oscillation is controlled by the values of L and C. L is varied only when changing from one band to another, as in an all-wave receiver, but C is ganged to the preselector tuning condensers and therefore is varied each time a new station is tuned in. Notice that in each case the rotor of C is grounded; this is done to simplify the construction of the ganged variable condenser of which C is one section.

The circuit in Fig. 7A employs inductive feedback (coil L_1). Automatic C bias is provided by C_g and R_g (R_g often will be found connected between grid and cathode, as indicated by the dotted lines), while C_p is a bypass condenser which keeps r.f. currents out of the plate supply. One way to couple energy into the first detector is to use a pick-up coil like L_M , which is inductively coupled to oscillator tank coil L.

▶ In the circuit of Fig. 7B the feedback is still inductive in nature, but for convenience in manufacture, coils L and L_1 have been combined into a single tapped coil. Resistor R_c and condenser C_c supply the customary form of automatic C bias. R_g and C_{pd} increase the C bias when the oscillator tends to develop excess tank circuit power (as it does at the higher frequencies), thus smoothing out or equalizing oscillator power variations. $C_{\rm pd}$ also serves as a padding condenser, the function of which will be considered later. This oscillator is shunt fed. R.F. current flows from the plate through condenser $C_{\rm k}$, while by-pass condenser ' $C_{\rm p}$ and resistor $R_{\rm f}$ keep r.f. currents out of the plate supply.

Hartley Oscillator. The form of Hartley oscillator circuit in which the plate is grounded (to permit grounding of the rotor of variable condenser C) is used in superheterodyne receivers; a typical circuit is given in Fig. 7C. Condenser C_p here grounds the plate for r.f. currents. Automatic C bias is used; there is no d.c. voltage across the variable condenser.

Service Hints. The oscillator stage in a superheterodyne rarely gives trouble but when it does, it usually goes dead and kills reception. The feedback is adjusted to give satisfactory results with average tubes; however, as the tubes age, its output may drop so that oscillation will not be sustained. Usually, a new tube will clear up the trouble.

Sometimes the grid resistor R_g will, be at the wrong value or will change in value with age. Check its resistance. If it has changed from the rated value, use a new resistance. The resistor can be reduced in value somewhat—the tube is just made to work harder and its life is slightly reduced. Too large a value will cause self-modulation or may even stop operation.

The Mixer-First Detector

The method used for mixing the output of the local oscillator with the incoming signal is highly important, for this mixing must take place with a minimum of reaction on the oscillator. There are two methods of mixing: electronic and external. Electronic mixing occurs in the electron stream within the mixer tube, while external mixing occurs in a grid or a plate circuit of the mixer tube (external or outside the tube). Today, the electronic method has largely replaced all others, particularly when the same tube is used as both the oscillator and the mixer-first detector.

However, we should study the methods of external mixing, as they have been widely used, and receivers using these systems, are still being serviced. Typical circuits for this purpose are shown in Fig. 8. One of the most widely used mixer connections of the external type is shown in Fig. 8A, where the oscillator output coil L_{M} is connected into the cathode lead of the mixer-first detector tube. R_{g} and C_{g} furnish automatic C bias voltage. The incoming signal acts directly upon the grid, changing its potential with respect to ground, while the oscillator signal changes the potential of the cathode with respect to ground. Both, therefore, affect the plate current which produces the desired mixing.

Another widely used mixer connection is that in Fig. 8B, where oscillator pickup coil $L_{\rm M}$ is in series with coil $L_{\rm P}$ (which feeds the incoming signal to the detector). The disadvantage of this circuit is that coil $L_{\rm M}$ is a part of the mixer input resonant circuit. Also, changes in $C_{\rm p}$ as the set is tuned may affect the oscillator.

▶ Some time ago, the oscillator tank

coil L and the preselector tuning coil L_p often were mutually coupled as in Fig. 8C, eliminating the need for-coil L_M . Both oscillator and preselector tank circuit coils were wound then on the same form and both often were placed in a shielded housing to prevent radiation. Today, however, this circuit is rarely encountered.

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Capacitive-resistive coupling between a high r.f. potential point on oscillator tank coil L and the control grid of the mixer tube occasionally is used, as shown in Fig. 8D. A small coupling capacity (from 10 to 100 mmfd.) is usually sufficient. Resistor R_{k} , of high ohmic value, is placed in series with this capacity to help isolate the circuits and prevent interactions as they are tuned.

Coil $L_{\rm M}$ in the oscillator circuits of Figs. 7A to 7C can be connected also in the screen grid, suppressor grid, or plate lead if the mixer-first detector tube is a pentode, as shown in Fig. 8E. The plate lead connection is very rare, however, for changes in plate voltage which ordinarily can be produced by oscillators have relatively little effect upon plate current. These connections are used when a large oscillator output voltage is available.

► A super-control pentode tube often is used as a mixer-first detector, for it closely approximates a square law detector on weak signals, giving distortionless frequency conversion, and it approximates a linear detector on large swings. It is necessary to use automatic volume control to make the operating point move automatically to a linear portion of the characteristic curve when a strong signal is obtained. When this is done, strong signals cannot swing the grid positive. There is

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some generation of harmonics with linear operation, but this is considerably less than would occur with a tube acting as a linear detector at all times.

COMBINATION OSCILLATOR-MIXER-FIRST DETECTOR

The introduction of multigrid tubes permitted a single tube to serve the function of oscillator, electron mixer, and first detector. The screen grid tetrode tube and the pentode tube were the first to be used for these combined functions.

A multi-function circuit using a pentode tube is shown in Fig. 9; this circuit was widely used in midget superheterodyne receivers. The oscillator section of the circuit is of the Meissner type, with coil L, variable condenser C, and padding condenser $C_{\rm pd}$ forming the oscillator tank circuit for tube VT. The plate of the tube is connected to (loaded by) the oscillator tank circuit L-C through two forms of coupling: 1. Inductive, through the mutual inductance between coils L and L_1 (condenser $C_{\rm LF}$ has negligible reactance at the frequencies generated by the oscillator); 2, Capacitive, through condenser C_{pd} , with the oscillator currents flowing from the plate through $C_{1,F_{1}}$, L_{1} and C_{pd} to ground. With this coupling arrangement the load on the oscillator tank coil is equalized over the entire band, resulting in more uniform oscillator output.

Feedback of the oscillator tank circuit voltage into the control grid circuit is obtained simply by connecting feedback coil $L_{\rm M}$ into the cathode circuit. This arrangement varies the potential between the cathode and ground. But since the grid circuit also eventually goes to the cathode, changing the potential of the cathode with respect to ground also changes the



FIG. 8. Methods which have been used for feeding the local oscillator signals into the mixerfirst detector of a superheterodyne. When $L_{\underline{w}}$ is shown, it is assumed to be coupled inductively to the oscillator tank circuit. In other cases, either direct inductive coupling or capacitive ceupling is used.

potential of the grid with respect to the cathode.

Having analyzed the oscillator and mixer functions of this circuit, we shall now see how it performs the duties of first detector. First of all, observe that the plate of the tube is fed (with d.c.) through coil $L_{\rm p}$, which serves as the primary of the first i.f. transformer and as the plate load coil for the first detector. The i.f. currents flowing out from the plate of tube VT thus flow through coil $L_{\rm p}$ and thence through bypass condenser C_{p} to ground. In addition, i.f. currents flow through the path formed by $C_{I.F.}$, L_1 and C_{pd} . The reactance of L_1 to i.f. currents is negligible, so $C_{I.F.}$ and C_{pd} together tune coil L_p to the i.f. value. (This circuit is completed from C_{pd} through the chassis and through $C_{\rm p}$ to $L_{\rm p}$.)

Automatic C bias is furnished by C_{z}

high $G_{\rm M}$ tubes. As a result, the bias may be too high for an average tube, and the original tube may stop oscillating when its emission weakens with age. If a receiver using this circuit fails to work when all parts are in good condition, try lowering the value of $R_{\rm g}$ by about one-third; this often will allow the original tube and, surely, all other new tubes to oscillate. If $R_{\rm g}$ is reduced too much, however, frequency conversion will not take place.

PENTAGRID CONVERTER TUBE

So much difficulty was experienced with screen grid and pentode tubes operating as combination oscillatordetectors that tube engineers developed a special tube which would permit independent biasing for optimum oscillator and optimum detector action. The pentagrid (five grid) tube was the re-



FIG. 9. Combination oscillator-mixer-first detector circuit using a pentode tube. If the suppressor grid is omitted, this circuit will apply to the screen grid tubes which were once widely used for the same purpose.

and R_s , the value of R_s being carefully selected to make tube VT function as both oscillator and detector. Optimum operation as an oscillator is more important than as a detector, because the detector characteristics are less critical.

Servicing Hint. Many of these circuits were built for specially selected, sult. Since this tube provides all the functions of the frequency converter section, it is called a *pentagrid converter tube*.

A practical frequency converter circuit employing a pentagrid converter tube is shown in Fig. 10. The oscillator triode section of the tube consists of the cathode, grid 1 (functioning as a control grid), and grid 2 (functioning as plate for the oscillator triode). (This second grid is called an "anode grid", to indicate its plate action.) Any desired form of oscillator circuit may be connected to these three electrodes; a standard tuned grid, plate coil feed-

the plate current, which is already varying at the oscillator frequency. Thus, the local signal and the incoming signal are mixed in the tube electron stream (this is called electron coupling or electronic mixing).

The C bias produced by R_d and C_d acts upon grid 4, thus controlling



FIG. 10. A practical frequency converter using a pentagrid converter tube.

back oscillator circuit is shown. Coil L, condenser C (rotor grounded) and padding condenser C_{pd} constitute the tank circuit of the oscillator, while coil L_1 in the circuit of grid 2 (the oscillator r.f. plate current back to the grid tank circuit to maintain oscillation. Condenser C_o and resistor R_o together provide automatic C bias for the oscillator. Since this bias is applied directly between grid 1 and cathode, it is independent of the automatic C bias created by C_d and R_d for the first detector.

The action of the oscillator sets up, just beyond the second grid, an *electron cloud* which serves as the *virtual cathode* for the other tube elements. Thus, the detector section is furnished an electron stream which is varying at the oscillator frequency. In the detector section of the tube, grid 4 acts as the control grid, grids 3 and δ as a screen grid, and the plate has its usual function. An incoming signal controls the flow of electrons from the virtual cathode to the plate. The shielding action of grids 3 and 5 is only partially effective. It is therefore necessary to choose a detector plate load (i.f. transformer primary) which will resonate sharply to the i.f., having at the same time such an L/C ratio that it will reject other signals. This is necessary in order to prevent i.f. harmonics and other undesired plate circuit signals from feeding back to grid 4 in the input circuit.

With oscillator and detector sections essentially coupled together only by the space cloud, it is possible to design the oscillator and detector sections of a pentagrid converter independently and secure optimum operation of each.

Disadvantages of Pentagrid Converter Tubes at Ultra-High Frequencies. Although the pentagrid converter tube gives excellent results at broadcast band frequencies and the lower short-wave frequencies, it is somewhat unsatisfactory at frequencies above



FIG. II. Another type of pentagrid converter.

about 10 megacycles. One reason for this is that despite the use of a screen grid, the oscillator section of the tube sets up a space charge which at high frequencies affects the input circuit (grid 4 in Fig. 10) directly. At ultrahigh frequencies, the oscillator circuit will cause regeneration when tank circuit L-C- C_{pd} is tuned below the frequency of the incoming signal, and will cause degeneration when this circuit is tuned above the frequency of the incoming signal. Since the oscillator of a super is ordinarily tuned above the incoming signal frequency, degeneration and consequent loss in signal strength occurs in the high-frequency band and at the higher frequencies in other bands. To avoid this degeneration, the manufacturers of some all-wave receivers make the oscillator tune below the incoming signal frequency on the highest frequency band.

Excessive oscillator frequency drift at frequencies above 10 megacycles is another trouble encountered in a pentagrid converter. The tank coil of the oscillator (L in Fig. 10) is affected by the a.c. plate resistance of the oscillator section of the tube (the resistance between grid 2 and cathode, which acts on the tank circuit through mutual inductance M). This resistance changes the oscillator frequency (as determined by the values of C, L and C_{pd}) a certain fixed percentage at all times. At low radio frequencies the error in oscillator frequency is so small as to be negligible, but at high radio frequencies the error is larger. The variation is relatively unimportant, however, for if the preselector is a little broad, it is only necessary to change the tuning dial setting slightly to bring in the station satisfactorily. More serious trouble occurs because of variations in carrier intensity (fading) when distant stations are being received. In any oscillator, detector, or combination oscillator-detector circuit, the a.c. plate resistance of the tube will vary with carrier intensity because variations in the carrier cause variations in the average space current of the tube. This varying a.c. plate resistance causes oscillator frequency drift, which can be very severe at the higher frequencies. \blacktriangleright Here is an example: Suppose that a carrier of varying strength changes the a.c. plate resistance enough to cause a maximum error of 0.1% in oscillator frequency. At 1000 kc. this percentage will give only a 1-kc. error in the i.f. beat, and the i.f. amplifier is almost always broad enough to offset such a small error. At 10 megacycles, however, this same percentage gives a 10kc. error in the i.f. beat, and clearly there will be severe cutting of side bands. Also, if the i.f. amplifier is highly selective, the signal may even fade out entirely unless the set is retuned. \blacktriangleright A special form of a pentagrid converter, which uses a 6SA7 tube, is shown in Fig. 11A. We can consider that the oscillator section of this circuit consists of the elements shown in Fig. 11B. As you can see, this is the kind of Hartley oscillator pictured in Fig. 7C, with the screen grid acting as

PENTAGRID MIXER-FIRST DETECTOR TUBE

To overcome the inherent shortcomings of the pentagrid converter at high frequencies, engineers designed a special pentagrid mixer-first detector tube which has a mixer or injector grid. When used with a separate oscillator tube, this pentagrid mixer-first detector tube is a better frequency converter because it has negligible frequency drift and negligible degeneration, even at the very high frequencies encountered in television receivers.



the plate or anode grid. In the detector section of Fig. 11A, grid 3 acts as the control grid, grids 2 and 4 form a screen for grid 3, and grid 5 is a suppressor. The cathode and plate, of course, also perform their usual functions for the detector section of the tube.

This form of pentagrid converter has somewhat greater sensitivity and less tendency toward oscillator frequency drift than has the circuit shown in Fig. 10. In addition, the 6SA7 tube is single-ended (has no grid cap) and so fits in better than other pentagrid converter tubes with the modern practice of keeping above-chassis wiring to a minimum. A practical circuit using this tube and capable of giving excellent frequency converter action well up into the very high frequencies is shown in Fig. 12. Tube VT is the special pentagrid mixer-first detector tube. Grid 1 is the control grid for the detector; grid 3 is the mixer grid; grids 2 and 4 are shielding grids for the mixer grid; while grid 5 is a suppressor grid.

Tube VT_1 is an ordinary triode tube connected into a conventional tunedgrid oscillator whose tank circuit consists of L, C, and C_{pd} . Condenser C_M provides capacitive coupling between the oscillator tank circuit and grid Sof VT, which injects the oscillator signal into the electron stream of the mixer tube. The stability of the oscillator is dependent only upon the design and construction of the oscillator stage and there can be no feedback of oscillator current to the preselector (the $L_{o}-C_{p}$ tank circuit).

► A triode-hexode tube has been designed which combines a triode oscillator and a hexode (6-element) mixerdetector in one envelope. A typical circuit in which this tube is used is shown in Fig. 13. As you can see, the oscillator section of this circuit uses a standard tuned grid oscillator. The grid of this triode section is connected within the tube to the first grid of the hexode section, which then becomes an injector grid and feeds the oscillator signal into the electron stream. Mixing occurs when the incoming signal is fed into grid 3 of the hexode section of the tube. Grids 2 and 4 of the hexode section merely act as shielding screens for grid 3.

This triode-hexode circuit has many of the advantages of the circuit in Fig. 12 for high-frequency operation and, in addition, permits the use of one less tube. Therefore, this circuit is being used more and more in all-wave receivers.

FREQUENCY CONVERTER RATINGS

When discussing the relative merits of different frequency converter systems, engineers need what might be called a "yardstick of comparison." What they really want to know is how well a frequency converter will change an r.f. signal voltage into an i.f. signal voltage—that is, frequency converter gain.

The important factor in determining



FIG. 13. A triode-hexode tube featuring electron mixing from the oscillator to the first detector.

the gain is the conversion transconductance (sometimes called translation conductance) of the frequency converter; it is equal to i.f. plate current $i_{\rm p}$ divided by r.f. input voltage $e_{\rm g}$ and is designated by the symbol Sc. Knowing Se, the gain of a frequency converter can be found by multiplying the value of S_c in mhos by the a.c. plate load impedance in ohms of the first detector circuit. For example, a 6A7 tube used with a reasonably powerful oscillator circuit will have a conversion transconductance of somewhere between 350 and 520 micromhos at ordinary frequencies, but this value will be reduced greatly at very high frequencies. A 6L7 pentagrid mixer-first detector used with a separate oscillator tube as in Fig. 12 will have a translation conductance of about 475 micromhos, this value changing little even at high frequencies. A commonly used value for the a.c. plate load impedance is 100,000 ohms. This load and a converter with an S_c of 400 micromhos (.0004 mhos) gives a gain of .0004 \times 100,000, or 40.

Oscillator-Preselector Tracking

Ganging of the preselector and the oscillator tuning condensers simplifies tuning of a receiver and eliminates repeat points. However, new adjustment problems are introduced when we attempt to make the oscillator operate above the frequency of the incoming signal by exactly the i.f. value at all times. The oscillator must follow or "track" the preselector, hence we call the alignment of the oscillator and preselector circuits a tracking adjustment. ▶ Let us see first why tracking adjustments are necessary. Turning to Fig. 14A, which shows both the preselector and oscillator tuning circuits, assume that the tuning condensers are set to minimum capacity, that L_p is exactly like L_{o} , and that C_{p} is exactly like C_{o} . Obviously, both circuits now will have the same resonant frequency. Since the oscillator circuit must tune to a higher frequency, the electrical value of either L_{\circ} or C_{\circ} must be reduced. It is not easy to reduce C_{o} , since its minimum capacity is due essentially to stray capacities between rotor and stator plates, so L_0 must be replaced with a coil having enough less turns to make the oscillator tune above the preselector by exactly the i.f. value when both condensers are at minimum-capacity positions.

But if, after making this adjustment, we set both condensers to their maximum-capacity values, we find that the oscillator is no longer exactly the i.f. value higher than the preselector. (The reason is that the numerical difference in frequency between the oscillator and the preselector depends upon the numerical values of both the inductances of the coils and the capacities of the condensers.) Therefore, when we change the numerical values of the capacities of the condensers we also change the numerical value of the difference in frequency.

Curves 1 and 2 in Fig. 14B show how the resonant frequencies of the oscillator and preselector circuits vary as the tuning dial (to which both condensers are ganged) is rotated from 100 to0 (100 corresponding to the minimum



FIG. 14. Preselector-oscillator tracking adjustments are necessary because the resonant frequencies of the two tank circuits (at A) do not differ by the same amount (the i.f. value) at each tuning dial setting. The discrepancy is shown by curves I and 2 (at B); curves I and 3 show the ideal relationship, which can be secured if the rotor plates of the tuning condensers are specially shaped as at C.
and 0 to the maximum capacity of the tuning condensers). As you can see, the resonant frequency of the oscillator decreases faster than does the resonant frequency of the preselector, so the i.f. beat frequency (the difference between the two) also decreases as the condenser is turned from minimum capacity to maximum capacity. There is need for an adjustment of some sort which will keep the beat frequency constant. We cannot reduce the number of turns on the oscillator coil, for the coil is correct as it now is for the high-frequency setting. Clearly, then, it is necessary to reduce the capacity of C_{\circ} for low-frequency settings.

 \blacktriangleright A constant i.f. beat frequency over the entire tuning range can be obtained by giving the oscillator rotor plates a different shape from those of the pre-



FIG. 15. Here are the adjusters used on the oscillator circuit when the tuning condenser does not have specially-cut plates.

selector. The exact shape required can be figured out by mathematics, but the general shapes will resemble those shown in Fig. 14C. Notice that as the frequency is lowered (by increasing the capacity), the oscillator condenser increases less in capacity than the preselector. Thus we obtain the results represented by curves 1 and 3 in Fig. 14B.

Condensers having specially cut rotor plates are expensive to build and are good only for one particular i.f. value and oscillator coil. Hence, they are clearly out of question for all-wave receivers, because here a different shape of rotor would be required for each band. For these reasons, all-wave receivers are built with ganged tuning condensers having all sections alike, and two trimmer condensers, known as the <u>low-frequency padder</u> and thehigh-frequency trimmer, are adjusted to make the preselector and the oscillator track each other.

Cut rotor plates are used extensively in broadcast band receivers, however. Also, they find wide use in auto radio receivers, as they keep to a minimum the number of trimmer condensers which might get out of adjustment because of vibration.

Low - Frequency Padder Condenser. When ganged tuning condensers are used, the oscillator condenser will have too high a capacity when its plates are completely meshed. (Remember that the oscillator coil has less inductance than the preselector coil.) This capacity can be lowered by inserting in the tank circuit, in series with C_{o} , a trimmer condenser which is called a low-frequency padder, a padder, or sometimes a lag condenser. This padding condenser can be adjusted to lower the tank circuit capacity just enough to give perfect alignment at the 0 (lowest frequency) point on the dial. The padding condenser shown as C_{pd} in Fig. 15 (and also in Figs. 9, 10, and 12) is considerably higher in capacity than the maximum capacity of C_o , so that its reactance is almost negligible (it acts as a short circuit) at the 100 or high-frequency setting of the dial.

However, at the low-frequency end, it acts in series with C_{\circ} and, as you know, condensers in series have a lower capacity than either one alone. Hence, the capacity in the circuit is reduced at the low-frequency end of the band, which provides the necessary tracking.

High - Frequency Trimmer Condenser. We have assumed thus far

that the exact i.f. value was produced at the 100 or minimum-capacity setting, but this seldom is the case in actual practice because, even with mass production methods, it is difficult to make coils and condensers with exactly the desired electrical values. Manufacturers of superheterodynes compensate for these errors by placing a small trimmer condenser (called the highfrequency trimmer condenser) in parallel with the oscillator tuning condenser. This, as well as the padder, must be adjusted to make the receiver function properly. Let us now consider how these tracking adjustments are made in a modern superheterodyne recciver.

► There are two alignment conditions: 1, where the receiver is completely out of alignment; and 2, where the alignment just needs "touching up" or adjusting for maximum response. Considering the first case, the following procedure would be used:

Assuming that the receiver is to be aligned for the 550-1500-kc. broadcast band, feed into the preselector input a 1000-kc. signal (supplied by a signal generator) and set the receiver tuning dial to its 1000-kc. division. Adjust the low-frequency padder C_{pd} for maximum receiver output; now you are getting exactly the correct i.f. beat for mid-dial settings. Change the signal input frequency to 1400 kc., set the receiver dial at 1400 kc., and now adjust high-frequency trimmer C_{T} for maximum receiver output. If there are two positions of $C_{\rm T}$ for which maximum output is obtained, choose the one at which C_{T} has minimum capacity; the other position is a repeat point. Now change the signal input frequency and the dial setting to 600 kc. and readjust the low-frequency padder for maximum output. It is necessary to repeat the adjustments at 1400 kc. and at 600

kc., since one trimmer has a slight effect upon the other. The final adjustment is made at 1400 kc. If the procedure has been correctly followed, the proper i.f. value will be obtained over the entire tuning range.

These alignment instructions apply to any other frequency band as well. High, medium, and low frequencies in each band are selected; the high-frequency trimmer and the low-frequency padder then must be adjusted to make the preselector and the oscillator track each other. The low-frequency padder is always adjusted for the low and midscale frequencies, while the high-frequency trimmer is adjusted at the highfrequency setting. This procedure is called a three-point track alignment adjustment. It makes the i.f. value correct for three points, insuring that the i.f. beat will not be off an appreciable amount at other dial settings.

On a few all-wave receivers, the padding condenser for some of the short-wave bands may be a fixed condenser rather than a trimmer. On such bands, there can be only a high-frequency adjustment, so only fair tracking is obtained.

▶ When the alignment is just to be "touched up" it is unnecessary to make the mid-frequency adjustment, as the high- and low-frequency points are the only ones likely to be off.

ALL-WAVE SUPERHETERODYNE RECEIVERS

Number of Bands Required. An all-wave superheterodyne receiver differs essentially from a broadcast band superheterodyne receiver only in that the all-wave receiver has one or more extra preselector and oscillator tuning circuits, which may be switched in as desired. Since it is difficult to tune a preselector and an oscillator over a range having a frequency ratio greater than about 3.3 to 1, it is necessary to

use a new set of coils in the preselector and oscillator circuits for each band of frequencies. If the broadcast band coils for a receiver cover the range from 540 kc. to 1780 kc. (1780 being 3.3 times 540), the next band will extend from 1780 kc. to 5800 kc. (3.3×1780) , and band No. 3 will extend from 5800 kc. to 19,100 kc. (these values are approximate). To secure this 3.3 to 1 range in each band, the variable condensers must have a low minimum capacity, stray lead connection capacity must be very low, and the coils must have a very low distributed capacity. When these requirements cannot be met because of chassis layout and design problems, or when it is desired to cover all frequencies from 540 kc. to about 22,000 kc., it is necessary to divide the entire frequency range into four bands and use four sets of coils.



FIG. 16. Preselector and frequency converter circuits of a four-band superheterodyne receiver which employs series coil switching.

Band Changing by Switching Series Coils. Changing the number of coils connected together in series is one way of changing the frequency range of a resonant circuit. Fig. 16 illustrates

how this is done in one practical fourband superheterodyne receiver. The four preselector coils are connected together in series, as also are the oscillator coils; when the band-change switches (operated by a single control knob) are set to band 1, only the uppermost coil is in each resonant cir-



FIG. 17. Preselector and frequency converter circuits of a two-band superheterodyne employing shunt coil switching.

cuit. Switching to band 2 adds another coil to each resonant circuit which lowers the resonant frequency by increasing the circuit inductance. Switching to band 3 adds another coil, and all four coils are used on band 4 (which ordinarily would be the broadcast band). Notice that a different low-frequency padder (x) is used in the oscillator tank circuit for each band. Only one highfrequency trimmer is used, so it is possible to make the high-frequency adjustment on only one band. (Usually this adjustment is made on the more popular broadcast band.) The preselector circuits must therefore be quite broad to compensate for poor tracking in other bands. Another undesirable feature of this series switching arrangement is the fact that the unused coils may absorb energy, making it necessary to use a special band-change switch which either disconnects or shorts out the unused coils.

Band Changing by Switching Shunt Coils. Band changing can be accomplished also by adding coils in parallel; Fig. 17 shows how this is done for a two-band receiver. Each oscillator coil has its own padder condenser to insure good low-frequency tracking. A trimmer condenser is used across one preselector coil to permit a high-frequency adjustment of the higher frequency band (band 1). Unused coils can still absorb energy in this circuit.

Quite often only one oscillator coil is used in two-band receivers, a harmonic of the oscillator being used for the higher frequency band. For example, if the oscillator range for the 550-1500-kc. broadcast band of a receiver having a 460-kc. i.f. value is 1010 kc. to 1960 kc., the second harmonic of the

Band Changing by Switching Complete Coils. Although the bandchanging circuits shown in Figs. 16 and 17 have been used in lower-priced receivers, perfect tracking on all bands is not obtainable, and energy absorption by unused coils can prove troublesome. A more practical and more widely used circuit is shown in Fig. 18 as it is applied to a four-band receiver. For each band there is a separate preselector coil with its high-frequency trimmer C, and a separate oscillator coil with its highfrequency trimmer C and low-frequency padder X. The change from one band to another is accomplished by a four-section (four-deck), four-point rotary switch. Since image interference is most objectionable in the broadcast band, an extra tuning circuit is inserted between the antenna and the main preselector when the band-change switch



FIG. 18. Preselector and frequency converter circuits of a four-band superheterodyne receiver in which band changing is accomplished by switching complete coils. One extra tuned circuit is used in the broadcast band (band-change switch set to No. 4) to give additional suppression of image interference.

oscillator will vary from 2020 kc. to 3920 kc., the correct variation for the 1560-kc. to 3460-kc. band. In this case band-changing is accomplished simply by tapping the preselector coil. is at position 4. With complete coils being switched, each band is electrically independent of the others and maximum operating efficiency is attained.

► An all-wave receiver circuit con-



FIG. 19. The preselector in this all-wave superheterodyne circuit uses two stages of r.f. amplification for the highest frequency band (band-change switch set to No. 1) and one stage for all other bands.

taining r.f. amplifier stages in the preselector is shown in Fig. 19. This arrangement gives better image rejection and improves the signal-to-noise ratio, thus lessening the effects of converter noise. Observe that one r.f. amplifier stage is used for the lower frequency bands (2, 3 and 4) but two stages are used for band 1 to offset the reduction in the sensitivity and selectivity of the receiver at high frequencies.

Thus, you can see that all-wave superheterodynes differ from broadcast band superheterodynes only in the sections ahead of the mixer-first detector. All-wave receiver circuits may appear complicated at first glance because of the band-changing switch and the extra parts, but connections can be traced quite easily if only one band is considered at a time. Various methods for shorting unused coils will be found; these often seem to complicate preselector circuit diagrams.

Combination Receivers. When the receiver is a combination AM-FM re-. ceiver, the widely different frequency bands require special considerations. It is possible to use a band switch and another set of coils, but many combination sets switch the entire preselector-oscillator-first detector section. This is desirable to keep down losses in the very high frequency FM bands, for this eliminates the use of long leads to band switches, and permits the use of tubes specially designed for each service. These combination receivers will be studied along with FM receivers.

The I. F. Amplifier

Most of the gain in a superheterodyne receiver is furnished by the i.f. amplifier. In fact, the very purpose of frequency conversion is to permit this section of the super to do its work. The i.f. amplifier also contributes most of the adjacent channel selectivity; that is, if the 1.f. amplifier is designed to pass all frequencies 5 kilocycles above and below the i.f. value, all stations producing i.f. beats which are outside this range will be tuned out or rejected by the i.f. amplifier.

CHOOSING THE I.F. VALUE

The receiver designer considers many things when he is choosing the i.f. value for a receiver. He knows that low i.f. values permit high i.f. gain and high selectivity, but require good preselector circuits to eliminate images and repeat spots. Likewise, he realizes that he can use a simple preselector if the i.f. value is high, and can still get reasonable gain by using coils which have pulverized iron cores. A high i.f. value is required if a wide band of side frequencies is to be passed. For a 10kc. band width, the i.f. should be at least 175 kc.; and for a 20-kc. band width, the i.f. value should be about 460 kc. The 6000-kc. band width used in television may require an i.f. value of at least 15,000 kc.

Should I.F. Values End in 5? It is common practice to choose an i.f. value which ends in 5, such as 175 kc., 265 kc., 465 kc., etc. The reason for this is interesting.

Harmonics of the i.f. signal, feeding back to the preselector (through the grid-to-plate capacity of the mixerfirst detector tube), produce very annoying squeals when stations are tuned in which have the same frequencies as these harmonics. For example, with an i.f. value of 175 kc., the fourth harmonic of the i.f. signal will be 700 kc. When a 700-kc. station is tuned in, the fourth harmonic of the i.f. frequency will beat with the 700-kc. carrier to produce an audio squeal frequency which will modulate the 700-kc. signal and, after frequency conversion, ride through the i.f. amplifier.

In the United States, stations in the broadcast band have frequencies ending in 10, such as 960 kc., 1440 kc., etc. By making the i.f. value end in 5, only its even harmonics can end in 10 and be equal to a broadcast frequency. For example, the harmonics of a 175-kc. i.f. are 350, 525, 700, 875, 1050, 1225, 1400, etc., of which 700, 1050 and 1400 represent station frequencies in the broadcast band; the harmonics of a 180-kc. i.f. are 360, 540, 720, 900, 1080, 1260 and 1440, of which all but the first value are equal to broadcast station frequencies. An i.f. value ending in 5 thus gives only half as many squealproducing points on the dial as an i.f. value ending in 10.

The higher the i.f. value, the fewer i.f. harmonics there will be in the broadcast band, which is the band suffering the most interference. This is another factor which makes 455 and 465 kc. the most popular i.f. values.

TYPICAL I.F. AMPLIFIER CIRCUITS

The most widely used i.f. amplifier is the twin resonant circuit shown in Fig. 20A. When C_1 and C_2 , the trimmer condensers, are tuned for maximum receiver output, a highly selective (sharp resonance curve) amplifier is obtained. When C_1 is tuned below and C_2 above the i.f. value, a rounded or even a double-peak response characteristic curve is obtained, but the gain is reduced considerably. In the latter case, the mutual inductance M must be of the correct value for band-pass results; this is a job for the designer. Wider band width can be obtained by shunting one of the coils with a 10,000- to 100,000-ohm resistor, but this gives even further reduction in gain.



FIG. 20. Typical i.f. circuits.

In any i.f. amplifier, the i.f. transformer in the plate circuit of the mixerfirst detector must be extremely selective, and must have a low L/C ratio in order to short-circuit the harmonics which are produced by detection which might otherwise feed back into the detector input. In addition, the coupling between the transformer coils must be loose; a copper screen is often placed between the two coils to reduce the coupling.

I.F. transformers, after the first one, can have higher L/C ratios and higher gain. Thus, these transformers may not all be alike—an important point when you are replacing one in service work.

▶ When high i.f. gain is desired with a minimum number of circuit parts, a single tuned i.f. circuit like that in Fig. 20B is used. Trimmer condenser C_1 may be placed across either L_1 or L_2 . You will find this circuit most often in midget supers. The gain is about double that of the type shown in Fig. 20A, butof course there is less selectivity.

High-Fidelity I.F. Amplifiers. In a high-fidelity receiver the resonance curves of the i.f. amplifier must be practically flat over the entire band of frequencies being transmitted, for this section provides most of the gain of the receiver. Other sections of the receiver must likewise pass all side frequencies without excessive attenuation (unless defects in one section are compensated by a correction in another section). Thus, if all frequencies up to 8500 cycles are desired, the i.f. amplifier must have a band width of 17 kc.

Variable Selectivity I.F. Amplifiers. High-fidelity receivers perform beautifully when tuned to strong local stations, but often give squeals and (sometimes called garbled specch "monkey chatter") when tuned to distant stations. This is because the response of the receiver is so broad that the carrier or side frequencies of an adjacent channel station can interfere with the desired carrier. The remedy obviously is a control which will permit the listener to choose between high selectivity and high fidelity (poor selectivity) characteristics. Three commonly used methods for obtaining variable selectivity in a superheterodyne are given in Fig. 21; each of these provides a continuously variable control, which allows the listener to reduce the fidelity just enough to stop the interference.

In the circuit of Fig. 21A the selectivity is changed by varying the mutual inductance M of the i.f. transformer (by mechanically varying the spacing between coils L_1 and L_2). With this arrangement, the trimmers are adjusted for peak response when M is a minimum (weak coupling). Now, when M is increased, the peak of the response curve flattens out, becoming flat or perhaps even double-humped. As you learned in an earlier lesson, increasing M has the effect of introducing capacity into one circuit while adding inductance to the other, which is exactly what is needed to get double peak response.

▶ Another scheme, which has the effect of shunting each resonant circuit with a resistance, is shown in Fig. 21B; this uses a third winding shunted by a variable resistor (R). The two resonant circuits originally are adjusted for peak response when R has a maximum resistance. Reducing R causes L_3 to absorb energy from both L_1 and L_2 , thus



FIG. 21. Three methods of securing continuously variable control of the selectivity of i.f. stages of a superheterodyne receiver are shown here. Although only one i.f. transformer is shown, each i.f. transformer in a receiver may be identically treated and the selectivity controls ganged together.

acting as a load. This widens the response curve, giving higher fidelity. A third method, shown in Fig. 21C, uses a third winding L_a which is connected in series with condenser C_3 and rheostat R. Resonant circuits C_1 - L_1 and C_2 - L_2 are tuned for maximum receiver output when R is set to maximum resistance. Then R is set to zero resistance and C_3 is adjusted for minimum output. Thereafter, the circuit L_3 - C_3 -R acts as a loading circuit. As L_3 and C_3 are resonant, the amount of energy absorbed depends on the setting of R. With R at a high resistance value, little energy is absorbed and maximum selectivity is obtained. When R is reduced toward zero, the other circuits are effectively loaded with resistance, reducing the selectivity and gain, and thereby broadening the over-all response. This circuit is just a variation on the one shown in Fig. 21B.

High-Low Selectivity Switches. In a variable-selectivity receiver the desired method of controlling selectivity must be incorporated in one or more i.f. circuits. To reduce costs, continuously variable selectivity is sometimes replaced with a high-low selectivity (low-high fidelity) change-over switch, as shown in Fig. 22A. Trimmer condensers C_1 and C_2 are adjusted for single peak response (high selectivity) when the switch is set to point 1. Moving the switch to point 2 makes coil section $L_{\rm M}$ common to both resonant circuits; this increases the mutual inductance of the coils. It also adds inductance to circuit L_1 - C_1 , and takes inductance away from circuit L_2 - C_2 , thereby increasing the band width (giving high fidelity).

Any scheme which will increase the resonant frequency of one circuit while decreasing the resonant frequency of the other circuit by an equal amount may be used to control selectivity. Another method of doing this is shown in Fig. 22B. When the selectivity switch is at point 1, normal peak adjustments exist. When it is at point 2, condenser

 C_3 is in series with C_1 and L_1 , increasing the resonant frequency of circuit C_1 - L_1 ; condenser C_4 is in shunt with condenser C_2 , lowering the resonant frequency of circuit L_2 - C_2 . Condensers C_3 and C_4 are of such capacities as to give the desired band width.

Comparison of I.F. Stages. It is interesting to compare the i.f. amplifiers of AM, FM and television receivers. Essentially, the i.f. amplifier delivers most of the amplification and selectivity in each of these receivers, and the basic design is similar. Fundamentally, they differ mostly in the degree of bandpassing and in the i.f. value used.

► Broadcast and so-called all wave AM receivers today use an i.f. value of 455 to 465 kc. (Broadcast receivers have used 1.f. values of 130, 175 and 260 kc. in the past.) In AM receivers intended for long-distance reception, the usual band-width is 10 kc. or less, so peak alignment is employed. The high-fidelity types are band-passed somewhat, out to a band width of 15 to 20 kc. It is important that the characteristic be relatively flat-topped so that there will be no discrimination against the modulation frequencies.

► The FM receiver has an i.f. value of from 2 to 4 mc., because the FM process requires a very wide band about 200 kc. Only a high i.f. value will permit such a wide band, and even here special design is necessary. It is not necessary that the response be flattopped as the limiter stage cuts off all peaks in amplitude, as you will learn when you study FM receivers more fully. Usually, the FM receiver will have a greater number of i.f. stages, with the last one or two being designed as limiters.

▶ In television receivers, the i.f. value is usually 13 mc., and the amplifier must have a band width of about 4.5 mc. to pass even *one* side-band. (Single



FIG. 22. Two methods for incorporating highlow selectivity switches in a superheterodyne receiver are shown here.

side-band transmission is used in television.) Here, special types of coupling are needed to give the necessary wide band. We will now briefly preview some of the requirements of a television receiver; there will be several lessons on these receivers further along in your Course.

Television Superheterodyne Receivers

Aside from the use of a cathode ray tube as an image reproducer, the superheterodyne receiver used to pick up sight (video) and sound (audio) television signals is much like an all-wave sound receiver. You have already studied in your course the basic circuits used in a television receiver, for no radically new principles are involved. However, we can present here only a preview of this subject. There are a number of lessons dealing with the special adaptations made on basic circuits to fit television requirements.

The transmission of a 525-line picture which is repeated completely thirty times each second calls for a video frequency range of about 4.5 mc. Partial single side-band transmission is used for television in this country, with the lower side frequencies being partially rejected at the transmitter. Both the sight and sound modulated r.f. carrier signals are radiated by a television -transmitter within a frequency channel 6 megacycles wide in the very high frequency band. Eighteen of these channels have been assigned to television; these channels are scattered over the frequency spectrum between 50 mc. and 300 mc.

In each television channel, the carrier center or resting frequency for the audio signal is .25 mc. below the highest channel frequency.* The carrier for the video signal is 4.5 mc. below the audio carrier, with side frequencies extending above and below this value in the manner shown in Fig. 23. At the receiver, both the sight and sound r.f. signals are picked up by the same antenna, and both are passed through the preselector to a frequency converter stage. A single local oscillator feeding into the frequency converter stage converts both the sight and sound signals and their associated side frequencies to lower i.f. values at which greater gain can be obtained and satis-



Courtesy Stewart-Warner Corp.



factory frequency response can be secured for all desired signal components.

As the sound and picture signals are on different carrier frequencies, they mix with the single oscillator frequency to produce two different i.f. values. The sight and sound signals thus separate at the output of the frequency converter stage. The sound portion of the television program goes into a sound

^{*}The sound portion of a television program is transmitted by means of frequency modulation; that is, the sound intensity variation is transmitted as r.f. frequency variations. The no-modulation frequency (corresponding to the carrier in a.m. systems) is referred to as the "resting frequency." Frequency modulation is discussed in another lesson.

i.f. channel having at least a 200-kc. pass band, then to the amplitude limiters and frequency modulation detectors (which are essential in f.m. receivers), from which the sound signal feeds into the audio amplifier and -loudspeakor. The sound channel includes provisions for rejecting picture signals.

y.



FIG. 23. Relationship between the audio carrier, the video carrier and the side frequencies in a typical 6-megacycle wide television channel. The frequency values specified below the diagram are for the 50-56 megacycle television channel.

► The sight carrier and its side frequencies pass into the video i.f. channel, which has a pass band almost 5.75 mc. wide. There are provisions in this channel for rejecting the audio carrier signal. The output of the video i.f. amplifier feeds through the video detector and video frequency amplifier to the television cathode-ray tube.

 \blacktriangleright A television picture signal includes impulses which indicate the end of a line and the end of a picture, along with the variations corresponding to detail along each line of the picture. Although the entire signal including the impulses is fed to the television cathode ray tube, only the video components corresponding to details in the scene vary the brightness of the spot on the screen. At some point ahead of the television cathode ray tube, a stage known as the clipper separates the video signals from the synchronizing impulses. After this. the line impulses are separated from the frame impulses. The separated impulses control oscillators which develop the voltages required to sweep the electron beam across the cathode ray tube screen in the proper manner to reproduce the original scene.

► The requirements for very wide pass bands-6 megacycles in the preselector, 5.75 megacycles in the i.f. amplifier, and up to about 4.5 mc. in the video amplifier stages-have given television receiver design engineers some knotty problems. These have been solved by using plate load resistances of low ohmic value and circuits having wide pass bands, but this results in low gain per stage. To compensate for this, special tubes with high mutual conductance values were developed for television receivers. These tubes also have low interelectrode capacities, to minimize attenuation of higher video frequencies.

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Lesson Questions

Be sure to number your Answer Sheet 23FR-3.

Place your Student Number on every Answer Sheet.

Send in your set of answers for this lesson immediately after you finish them, as instructed in the Study Schedule. This will give you the greatest possible benefit from our speedy personal grading service.

- 1. Why must the resonant load in the plate circuit of the first detector be highly selective? In order to tune any signia component from the others
- 2. What two radio frequency signals in a superheterodyne receiver are combined by the frequency conversion process to produce the i.f. carrier?
- 3. In addition to giving single-dial tuning, what advantage is secured by signal ganging the oscillator and preselector tuning condensers?
- 4. When an interfering signal is heard along with the desired signal, and the frequency of this interfering signal is above the frequency of the desired signal by twice the i.f. value of the receiver, what type of interference is present?
- 5. How would you eliminate code interference which is heard at all settings of the tuning dial in a superheterodyne receiver? Page 10
- 6. Why is it desirable to have a stage or two of r.f. amplification ahead of the frequency converter section in a superheterodyne receiver?
 - 7. Give two reasons the the pentagrid mixer-first detector tube, when used with a separate oscillator tube, is a better frequency converter than a single pentagrid converter tube.
 - 8. What two trimmer condensers must be adjusted in order to make the preselector and the oscillator track each other?
- Low bugnesses Padder of High Inducing frammer,
 In what way, essentially, does an all-wave superheterodyne receiver differ from a broadcast band superheterodyne receiver? Page 27
- 10. What section of a superheterodyne receiver provides most of the adjacent channel selectivity?

World Radio Histor

SMILE

When you give a smile, you give something which is priceless yet costs nothing. Nobody can buy, beg, borrow or steal your smile, because it is of no value unless yon give it away in friendly greeting.

A smile takes but a moment, but its effects sometimes last forever. A smile creates happiness among friends, brings sunshine to the sad, and promotes valuable good will in business.

You don't feel like smiling? Then force yourself to smile. Whistle, hum a tune or sing softly—act as if you were already happy, and the smile will come. A happy smile comes from happy thoughts, not outward conditions.

It isn't what you are or where you are or what you are doing that makes you happy or unhappy—*it's* what you think about it. You'll find just as many happy faces among Chinese coolies sweating in the rice paddies for ten cents a day as you will among any similar-size group of business presidents in this country.

Smile!

J. E. SMITH

TONE CONTROL, **VOLUME EXPANSION AND NOISE LIMITING CIRCUITS** Saind W. Camp

24FR-1

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STUDY SCHEDULE NO. 24

For each study step, read the assigned pages first at your usual speed, then reread slowly one or more times. Finish with one quick reading to fix the important facts firmly in your mind, then answer the Lesson Questions specified for that step. Study each other step in this same way.

1. Satisfying the Human Ear
2. Simple Tone Controls; Special Bass and Treble Tone Controls
3. Automatic Tone Controls; Unique Tone Controls Pages 9-15 Condenser-Type A.B.C. Circuit; Series-Resonant A.B.C. Circuit; Separate Bass and Treble Controls; Motorola Acoustinator; G-E Tone Monitor. An- swer Lesson Questions 5 and 6.
A. Volume Expansion Circuits
5. The Noise Problem; Noise-Reducing Tone Controls; Noise Impulse-Silencing Circuits
6. Inter-Carrier Noise Suppression Circuits Pages 24-28 Biased Demodulator Circuit; A.F. Amplifier Blocking Circuit; I.F. and R.F. Amplifier Blocking Circuit. Answer Lesson Questions 9 and 10.
7. Mail Your Answers for this Lesson to N.R.I. for Grading.
8. Start Studying the Next Lesson.

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World Radio History

Tone Control, Volume Expansion and Noise Limiting Circuits

Satisfying the Human Ear

THE manufacturers and operators of sound broadcast transmitters have one objective in common with the manufacturers of radio receivers-to make the receiver owner feel that he is listening to the actual studio performance rather than to a loudspeaker. This is by no means a simple task, and as yet has not been completely achieved even in the most expensive radio receivers. Radio engineers are doing everything within their power to improve the faithfulness of reproduction, and each year brings them another step closer towards their goal of perfection.

In this text-book we shall consider the various factors which make it difficult to achieve complete perfection, and study the methods used to satisfy the human ear under different conditions.

Distortion. One important requirement for faithful reproduction is freedom from all types of distortion. All of the sound frequencies which are present in the original performance must be present in the reproduced program and must seem to have the same loudness relation to each other as in the original performance; furthermore, there must be no frequencies in the reproduced program which did not originally exist (that is, there must be negligible amplitude distortion and negligible noise originating in the transmitting and receiving apparatus).

Noise. Broadcast transmitters can be, and in most cases are designed to radiate negligible noise. Radio receivers can likewise be so designed that noise originating in them is negligible. It is noise which originates outside of the transmitting and receiving apparatus which requires special attention. Man-made interference noise is the chief offender in this respect; for this reason every serviceman should be familiar with the use of noise-reducing antennas for receivers and the application of noise filters to trouble-making electrical devices, all of which are taken up elsewhere in this Course.

Atmospheric noises, which enter the receiver along with the carrier of the desired station, cannot be entirely eliminated, but their annoving effects can be reduced considerably. You will learn that there are three practical types of circuits used for this purpose: 1, Tone Control circuits, which cut out high-frequency sound components, thereby eliminating predominating noise signals at a sacrifice of fidelity; 2, noise impulse-silencing circuits. which temporarily cut out all signals for the duration of sharp noise pulses which are stronger than the program signals; 3, inter-carrier noise suppression circuits, which cut out all signals. including noise, whenever the station signal is so weak that it is drowned out by noise signals. (Increasing the amount of power radiated by a transmitter, in order that the desired signal can better override noise signals, is a fourth solution, but transmitting power is definitely limited by law in order to prevent interference between stations.) Typical examples of the ingenious circuits which have been developed to accomplish these results will be studied in this lesson.

Peculiarities of the Human Ear. The characteristics of the human ear have a lot to do with radio receiver design. Possibly you have noticed that orchestra music which sounds fine at a moderate distance away seems highpitched-(lacking in_low frequencies) when you walk a considerable distance away.

The same phenomenon occurs during the fading of a radio program; this can most readily be noticed when a foreign short-wave program is tuned in. At a normal listening level of loudness the music will sound fine, but as the signal fades out, the music seems to become high-pitched or squeaky; as signal strength comes up to normal again, the music gradually becomes more mellow and pleasing.

The reason for this puzzling effect is simply that our ears hear low frequency or bass notes better at medium and high loudness levels than at low loudness levels. When the volume of a radio program drops because of fading, the bass notes seem to be cut far more than the higher frequencies.

Since radio receivers in homes are ordinarily operated at a loudness level considerably lower than that of the original performance, we have the queer situation that even a perfect radio receiver (one having a flat response over the entire audio range) would sound unsatisfactory whenever its output volume differed from that of the original program.

Characteristic Curves for the Average Human Ear. The peculiar hearing characteristics of the human ear have been carefully investigated. The results of tests on thousands of persons have been combined in the graph in Fig. 1, which tells how the average ear responds to various frequencies at different loudness levels.

A few words of explanation as to how data for this graph was secured will help you to read its story. First of all, a 1,000-cycle sound was varied in loudness, while various persons listened, until an average loudness level which could just barely be heard was determined. This was called the 0 (zero) db-loudness level. - The actual _ intensity of this 1,000-cycle sound was then determined with a sound-measuring instrument, and the meter scale on the instrument was made to indicate 0 db intensity level for this condition. Now the same test was repeated for other frequencies, and the intensity level in db at which the sound could just barely be heard was measured with the instrument. Thus, at 500 cycles the average meter reading was +6 db; at 100 cycles it read 38 db, and at 30 cycles the reading was +63 db, etc. Each value was plotted on the graph, and a smooth curve was drawn through all the points to give the 0 db loudness level curve, also known as the threshold of hearing curve. This curve shows clearly that intensity levels considerably greater than 0 db are required below 1,000 cycles in order for sound to be heard, and also shows that the human ear hears best at about 4,-000 cycles per second.

The intensity level at 1,000 cycles was then raised 10 db to give a loudness level of 10 db at this frequency, and the intensity level at each other frequency which seemed to give this same loudness was determined. This gave data for the 10 db loudness level curve. The same procedure was repeated at 10 db intervals to secure data for the remaining curves in Fig. 1.

It was found that when the loudness level was made higher than 120 db, the sounds were actually felt by the persons taking the test, the sensation being that of pain in the ears in some instances; the 120 db loudness level curve is therefore known as the <u>thres-</u> hold of feeling. How to Use the Ear Characteristics Graph. Suppose we want to determine the intensity level required for a certain loudness, say 20 db, at a given frequency such as 200 cycles. We simply locate this frequency on the horizontal scale (point A), trace directly upward from it to the point where that frequency line intersects the 20 db loudness curve (point B), and then trace horizontally to the vertical reference scale at the left, where we read +40db (point C) as the intensity level required in db for that loudness. In other words, for a 20 db loudness level microphone which picks up all sound frequencies equally well.

We can draw these general conclusions from the curves in Fig. 1: At low loudness levels (from 20 to 40 db), such as those ordinarily produced by a radio receiver loud-speaker in the home, the human ear is most sensitive to sounds in the middle frequency range from 500 cycles to 5,000 cycles, and has difficulty in hearing sounds below and above this frequency range.

As we raise, the *loudness level* of sound, such as by turning up the volume control on the receiver, the sensi-



FIG. 1. This standard reference graph gives the average hearing characteristics (frequency response) of the human ear at various loudness levels; in other words, it tells what intensity level will be required for a given loudness level at a given frequency.

our ear requires 20 db more sound power at 200 cycles than at 1,000 cycles. At a 90 db loudness level, however, the curve is practically horizontal between these two frequencies, indicating that our ear can hear both frequencies equally well when they are this loud.

Bear in mind that <u>loudness level</u> is a human response to sound, and not something which can be measured with instruments. <u>Intensity level</u> is the actual measured level of sound as measured by instruments connected to a tivity of the ear to low frequencies improves rapidly, so that at a loudness level of 100 db we can readily hear all sound frequencies below 10,000 cycles equally well. The lower the volume control setting on a radio receiver having an ordinary volume control, then, the fewer low or bass notes there will be insofar as our ears are concerned; to a lesser extent this statement also applies to the extreme high frequencies.

To get true high-fidelity reproduction at any desired loudness level, then, it is necessary to compensate for the shortcomings of the human ear by boosting the response of a receiver at the low and high frequencies the correct amount for each loudness level. Boosting of the lows (raising of the bass notes) is far more important than raising the high-frequency or treble sounds.

Tone Controls. Many receivers are provided with separate bass and treble tone controls, which allow the listener to adjust the response of the receiver to secure faithful reproduction or to suit his particular taste at any desired loudness level. Oftentimes, however, the radio engineer makes these tone controls entirely automatic in operation, so that as volume is reduced, the low and the high frequency notes are automatically strengthened with respect to the middle frequencies. The amount of tone compensation provided at each loudness level is governed by the average reaction of the human ear to sound; receiver design engineers continually refer to the standard curves in Fig. 1 for this purpose. Automatic tone control is particularly appreciated by those listeners who do not have the ability to judge the correct adjustments of the bass and treble controls, or who do not want to take the time required to make these adjustments. Both manual and automatic tone controls will be taken up in this lesson.

Volume Compression. Of course, no radio program is broadcast at a constant volume level from instant to instant. For example, a symphony orchestra may have a variation in intensity of as much as 100 db from the weakest to the strongest passages. If the transmitter were adjusted to handle the loudest passages without overmodulation, then the weakest passages would be completely drowned out by transmitter circuit noise, and

the average modulation percentage will be too low for economical operation; on the other hand, if the transmitter is set to keep the lowest passages well above the transmitter circuit noise level, then severe overmodulation and distortion would occur on loud passages. A maximum variation of 40 db is about the most a transmitter can handle, although a range of 30 db seems to be preferred in most transmitters. If the signal level is allowed to rise too high, over-modulation and distortion occur; oftentimes this also results in overloading of the transmitter and opening of circuit breakers. throwing the station temporarily off the air. It is the duty of the studio operator who monitors a broadcast to reduce the audio gain of the transmitter for loud passages and boost the gain for weak passages, thereby compressing the sound into the allowable volume range. Automatic volume limiting systems are also employed for this purpose; these compress just before over-modulation occurs, by cutting down gain.

Volume Expansion. The compression of the original sound into a limited volume range will reduce the quality of the transmitted program. In a number of radio receivers special volumeexpanding circuits are used to counteract this compression at high levels and increase the range of volume somewhat. Full expansion is ordinarily not possible because of the inability of the amplifier in the receiver to furnish the power required for the loudest sounds produced in the studio and because only a part (or none) of the compression occurring at the transmitter is automatic, but there is an appreciable improvement in the quality of reproduction when a volume-expanding circuit is incorporated in a receiver.

Sounds are compressed in much the

same way when phonograph records are made, and many of the better phonograph amplifiers have volumeexpanding circuits. Because the high level compression usually is automatic in recording studios, the expansion can be made to correspond almost exactly, and very good fidelity can therefore be secured. Several different automatic volume expansion circuits will be studied.

Directional Effects. When we listen to an actual orchestra, our ears are able to determine from which direction each particular sound comes: thus we may note that the drums are to the right of the conductor, while the bass horns are at his left. When all these sounds are reproduced by a single loud-speaker at the receiving point, however, we lose this effect completely, and all of the sounds appear to come from the loudspeaker. With our present system of broadcasting, it is impractical to correct this situation, although it has been done already on an experimental basis.

In one test broadcast, two microphones were located a definite distance apart in the studio, and the sounds picked up by each were independently amplified and broadcast over two separate transmitters. At the receiving location were two separate receivers and loudspeakers, with the loudspeakers located the same distance apart as the microphones in the studio. As a result, the actual directional characteristics of the orchestra were duplicated.

Curiously enough, the lack of directional qualities in a radio program in no way destroys the entertainment value of the program for the average listener. We have become accustomed to radio transmission as it is, and consequently the tremendous expense involved in complete duplicate transmitting and receiving systems for each broadcast is unwarranted at the present time.

Simple Tone Controls

Any device which, when introduced into a radio receiving circuit, serves to reduce or remove the higher-frequency sound signals will have the apparent effect of boosting the low and medium frequencies. A condenser is one simple device for doing this, and is therefore widely used in simple tone controls. It is customary to make the insertion of one or more condensers in the circuit for tone control purposes optional, so the listener can select the most satisfactory tone control position for each loudness level.

When a tone control condenser is connected between the grid and the cathode or chassis of an audio amplifier stage or between the plate and the cathode or chassis, the high reactance of this condenser at low frequencies results in negligible by-passing of the signal, but at medium and high frequencies the reactance of the condenser is so low that there is appreciable signal by-passing. A simple condenser tone control gives an apparent boost in bass or low-frequency signals because the signal which it feeds to the next stage has a higher proportion of bass notes (with respect to' medium and treble notes) than the signal fed into the tone control.

Simple Four-Position Tone Control. One simple and widely used tone control arrangement is that shown in Fig. 2A, where selector switch SW serves as the tone control. When this switch is set at point 1, no shunting or by-passing of the signal occurs, and hence all frequencies in the sound signal are fed to $C_{\rm K}$ without attenuation. As the switch is set to points 2, 3 and 4, increasingly greater capacities are placed between the plate and ground, and increasing amounts of the middle and high frequencies are shunted to ground without passing through the load resistor $R_{\rm L}$ for transfer to the following stage.

- In inexpensive receivers there are generally only two positions of the tone control, and only one condenser; in the first position of the tone control switch, which might be labeled "BRIGHT," no condenser is in the circuit and all frequencies are passed uniformly. In the second position of the switch, often labeled "MELLOW," the condenser is in the circuit and the medium and higher frequencies are cut that they will not sound any louder to the human ear than the low and medium frequency signals. It must be remembered that the curves in Fig. 1 were secured by averaging the hearing characteristics of a great many persons. Individuals may therefore deviate considerably from these curves.

Continuously Variable Tone Control. In the circuit of Fig. 2A, the tone control must be adjusted in steps, with the result that proper correction of tone will not be possible at certain loudness levels in between these steps. For this reason many people prefer a continuously variable tone control like



FIG. 2. Simple manual tone control circuits which utilize the signal by-passing action of a condenser.

down, giving a bass-boosting effect. -Receivers with three-position tone controls, labeled "BRILLIANT," "BRIGHT" and "MELLOW," would use two condensers.

What Is the Best Tone Control Setting? It is incorrect to say that highest, fidelity is always obtained at one particular position of the tone control. When the receiver is turned to maximum volume, giving a high level of loudness, that position in which there is no condenser in the circuit to by-pass the higher frequencies will give the best fidelity if the receiver normally has a flat response. On the other hand, when the receiver volume is turned down, it may be necessary to place the highest-capacity tone control condenser in the circuit in order to reduce the highest frequency signals enough so

that shown in Fig. 2B. Here a variable resistor is placed in series with a single tone control condenser. When this resistor is set at zero resistance, only the condenser is connected between the plate of the tube and ground, and we have maximum reduction of the high frequencies (the low and medium frequencies will then come through considerably stronger than the high frequencies). Increasing the resistance increases the net impedance of this tone control path to ground, thus reducing the by-passing effects on the high frequencies. When the maximum tone control resistance of 100,000 ohms is in the circuit, practically no signal frequencies take the path through the tone control circuit and there is no cutting of high frequencies.

A similar continuously variable tone

control arrangement is shown in Fig. 2C as it would be connected between the grid and chassis of an audio amplified stage. The action of this circuit is identical to that in Fig. 2B; a lower value of capacity and a higher value of resistance are required for the grid connection, as you can see by comparing the values specified in Fig. 2C with those in Fig. 2B. It is important to realize that an appreciable amount of attenuation or cutting of high frequencies can be achieved only if the impedance of the tone control circuit is considerably lower than the impedance of the load which it shunts. Since the grid-cathode path of the tube in Fig. 2C is a high-impedance load for the secondary winding, C_{T} and R_{T} have high ohmic values.

How Tone Controls Reduce Noise. Simple tone controls are widely used in all-wave receivers, and are especially valuable when listening to short-wave programs coming from far distant transmitters. The signals from these distant stations are generally accompanied by a great deal of noise; this is because they are received at such low levels that receiver gain must be increased considerably, and noise then becomes more noticeable. It is a known fact that in these noise signals the higher sound frequencies predominate; when the tone control is set to cut down high frequencies, the received signal becomes considerably less noisy. You can easily verify this for yourself by trying the tone control on an allwave receiver while tuned to a European station; note how the noise is reduced when the tone control is set for MELLOW or BASS reception.

Tone Controls May Be Connected Anywhere in the A.F. Amplifier. There appears to be no one preferred position for the tone control in an audio amplifier. Sometimes this control is

connected in the output circuit of a diode detector, right at the input of the audio amplifier, as shown in Fig. 3. Potentiometer R_3 serves as the volume control, with the setting of its movable arm determining the amount of A.F. voltage which is fed through blocking condenser C_4 to the grid of the first audio stage. Tone control potentiometer R_2 is shunted across R_3 , and the movable contact of R_2 is grounded through condenser C_a . Moving the tone control from position 1 to position 2 gradually increases the shunting effect of C_3 , and therefore increases the cutting or by-passing of high-frequency signals.

A similar tone control arrangement could be used in Fig. 2B by replacing



FIG. 3. Simple manual tone control circuit acting on the load of a diode second detector; the parts values are those used in the Silvertone Model 1825A superheterodyne receiver.

resistor $R_{\rm L}$ with a potentiometer and connecting the movable tap of the potentiometer to the chassis through a .02 mfd. condenser; $R_{\rm T}$ and $C_{\rm T}$ would then be removed. Likewise in Fig. 2C the secondary of the audio transformer could be shunted with a .5 to 1 megohm potentiometer, with its movable contact grounded through a condenser, in place of the $R_{\rm T}$ - $C_{\rm T}$ tone control arrar.gement shown.

How Tone Controls Affect A.F. Response Curves. The effect of a typical four-position tone control upon the over-all audio frequency response of a

radio receiver is clearly shown by the graph in Fig. 4. Curve 1 is for the tone control setting where there is no shunt capacity to cut down the higher frequencies; this curve therefore represents the normal frequency response of the receiver, indicating the relative output intensity level at each sound frequency. Curves 2, 3 and 4 show how this response is changed as increasingly greater shunt capacity is inserted in the circuit. These curves show clearly how a tone control cuts down the medium and high frequencies in the audio range without affecting the low or bass notes. With variable tone controls like those in Figs. 2B, 2C



FIG. 4. This graph shows the effect of a fourposition tone control upon the over-all frequency response of a typical radio receiver.

and 3, we could have any number of different response curves between 1 and 4, depending upon the control setting.

If the volume level of a receiver is raised to a high value when the tone control setting corresponds to curve 4 in Fig. 4, bass notes will be amplified excessively for the requirements of the human ear, and the over-correction will be readily noticed by the listener as a boomy effect. Some people actually prefer a very strong bass response and will purposely set the tone control for this result, even though they thus destroy the fidelity or quality of reception.

Special Bass and Treble Tone Controls

In the more expensive radio receivers you will often find two separate tone controls, one for changing the lowfrequency or bass response and the other for changing the high frequency or treble response. Let us first consider bass controls.

Parallel Resonant Bass Control Circuit. Resonant circuits are widely used to control the bass response of a receiver; a typical example is that in Fig. 5A, where condenser C and iron-core inductance L form a parallel resonant circuit which has a resonant frequency of about 50 cycles (C will ordinarily be about .1 mfd., and L about 100 henrys). At resonance this circuit acts as a high resistance which, being in series with resistor R (about 10,000 ohms), will result in a high plate load impedance. The amplification of the stage is therefore increased in the low frequency or bass range around 50 cycles.

This parallel resonant circuit acts like a reactance of low ohmic value at frequencies higher than 50 cycles, under which condition resistor R governs the plate load impedance. The greater the resistance of the resonant circuit at resonance, the greater will be the boost in low frequency response; this resonant resistance can be varied over a wide range by adjusting the setting of potentiometer $R_{\rm B}$. When all of $R_{\rm B}$ is in the circuit, the resonant resistance is quite low and there is little or no boosting of bass response. Maximum bass boosting is secured when all of $R_{\rm B}$ is shorted out.

Audio Input-Shunting Bass Control Circuit. Another widely used bass tone control circuit is shown in Fig. 5B. Inductance L, potentiometer R and condenser C form the bass tone control circuit, which in this case is not a resonant circuit.

We can neglect the effect of C_{κ} , as its reactance will be small in comparison to the reactance of L at any frequency or in comparison to the resistance of R. Assume for the moment that C is omitted; clearly we now have a voltage divider consisting of R and L, with the reactance between point Pand ground determining the voltage output. Since the reactance of L at low frequencies is quite small in comparison to the resistance of R, the bass voltage will be developed essentially across R (between points 1 and 2). As potentiometer contact P is moved from point 2 to 1, less and less of this bass voltage is fed to the grid of the next tube. Since the audio system is intentionally made to have a high bass response, position 2 of the movable tap is for high bass response and position 1 is for low bass response.

Although moving contact P from point 2 to point 1 decreases the amount of bass forwarded to the following tube, it also decreases to some extent the medium and treble frequencies. Condenser C in Fig. 5B is used to bypass medium and high frequency signal components around the upper section of R, and consequently the medium and treble notes are passed by the tone control circuit without appreciable attenuation regardless of the setting of P. This bass control is definitely not a volume control when condenser C is used; if C is omitted, however, volume control action will be present along with bass control.

Treble Controls. The resonant tone control circuit in Fig. 5A can also be made to serve as a treble control if the values of L and C are chosen to provide resonance at a high audio frequency value. With this arrangement, varying the setting of potentiometer $R_{\rm B}$ will vary the high frequency response from its normal value to a value considerably above normal.

Use of a Conventional Tone Control as Treble Control. A more widely used procedure for securing treble control is that wherein the audio amplifier is designed to have a sharp rise in response at the higher frequencies, and one of the conventional tone control circuits shown in Figs. 2A, 2B, 2C and 3 is used to cut down this high frequency or treble response as desired.

Automatic Tone Controls

The purpose of a tone control, as



FIG. 5. Typical bass control circuits; that at A can also be made to serve as a treble control, by making L-C resonate at a high audio frequency. That at B is used in a General Electric all-wave receiver.

you already know, is to produce the effect of boosting the extreme low and extreme high-frequency response of the receiver to compensate for the inability of the human ear to hear these frequencies as well as the medium frequencies at low levels of loudness. The same results can of course be accomplished by reducing the level of the middle frequencies more than that of the extreme frequencies when reducing the volume. It is possible to secure this action automatically as the volume control is adjusted by the listener.

Engineers differ as to the extent to which automatic tone control should correct for the shortcomings of the human ear at low levels of loudness. Some feel that only the bass should be reinforced (automatic bass compensation), while others prefer to reinforce both the treble and the bass (automatic tone compensation) to secure more nearly perfect compensation for the drop in sensitivity of the human ear under the conditions in question.

Automatic bass compensation (often abbreviated as A.B.C.) could be secured with a tone control of the type shown in Fig. S if we mounted this tone control potentiometer on the same shaft as the volume control, so that when volume was reduced, the medium and high frequencies would be cut down proportionately in order to satisfy the human ear. While this could be made to work perfectly satisfactorily, there is a much simpler way of accomplishing the same results.

Condenser-Type A.B.C. Circuit. A practical automatic bass compensation circuit is shown in Fig. 6A; although quite simple, it is entirely effective. The two terminals of this circuit, x and y, are connected either across the diode load resistor or across the output of some audio stage in the receiver. Volume control potentiometer R_1 is tapped at point 2, which need not necessarily be its mid-point. Condenser Cin series with resistor R between points 2 and 3 serves to attenuate (cut down) the middle and high frequencies at low-volume settings of the control, thus providing automatic bass compensation. Whenever you encounter a circuit like this, where the volume control has a fixed tap which is connected to ground through a condenser and a resistor, you can immediately identify it as an automatic bass compensation circuit.

Consider the circuit in Fig. 6A from the viewpoint of a voltage divider. If C and R were not present, the voltage E at any frequency would depend solely upon the position of contact arm P. With C and R connected as shown. and with the reactance of C decreasing with increases in frequency, the impedance between points 2 and 3 decreases as the frequency increases. The lower this impedance becomes, the more A.F. voltage there will be across section 1-2 of the voltage divider; in fact, at very high audio frequencies almost all of the available signal voltage is developed across points 1 and 2.

When P is at point 1, all of the A.F. voltage developed across the voltage divider is fed to the grid of the next stage regardless of its frequency. If Pis placed at point 2, only the voltage developed between points 2 and 3 is fed to the grid of the next stage; the reactance between these points decreases at the higher frequencies and consequently there are fewer high-frequency components than normal in the output voltage E when P is at 2. We thus have the bass-boosting effect required for automatic bass compensation. At any position of P between points 1 and 2, the output voltage Edepends upon the ratio of the impedance between P and S to the total impedance between points 1 and 3, and the amount of automatic bass compensation increases from zero (when P is at 1) to its maximum value when P is at 2. When P is moved below 2(towards 3), the amount of bass compensation remains the same as for point 2 and all frequency components are reduced uniformly as in normal volume control action.

Occasionally you may find an additional condenser and resistor connected in series between point 4 and the chassis, and another condenser and resistor between point 5 and the chassis; this is done to provide increasingly greater bass compensation as contact arm P is moved from 2 to 3.

Series-Resonant A.B.C. Circuit. Even greater attenuation or cutting of the high and middle frequencies for automatic bass compensation purposes is possible with a series resonant circuit arrangement as shown in Fig. 6B. Coil L and condenser C in series are adjusted to be broadly resonant to the middle and high frequencies; as a result, the effects of their reactances cancel at these frequencies, leaving only resistance R and the coil resistance in shunt with section 2-3 of

potentiometer R_1 ; resistance R prevents complete cutting of signals at the resonant frequency. At low frequencies, however, the reactance of Cis so high that the shunting effect of the series resonant circuit can be neglected. As contact arm P is moved from point 1 to point 2, lowering the volume, the amount of cutting of medium and high frequencies gradually increases to a maximum value when Pis at 2. Further reductions in volume (by moving P below point 2) give no further increase in bass compensation unless an additional series resonant circuit or condenser-resistor circuit is connected between point 4 and ground. Very often resistor R is omitted and a coil which has a sufficiently high resistance to give the desired broad tuning over the middle and high frequency range is used. If the L-C circuit is broadly resonant only for the medium frequencies, only these medium frequencies will cut down at low volume control settings, and automatic tone control (bass and treble compensation) is secured.

Unique Tone Control Circuits

Separate Bass and Treble Channels. Some manufacturers of radio receivers provide one path in the audio amplifier for low-frequency signals and another entirely separate path for highfrequency signals; this permits amplification of these signals independently



FIG. 6. Two practical automatic bass compensation circuits.

and gives a definite control over their relation to each other.

A simple circuit of this nature is shown in Fig. 7. The audio signal output of the first A.F. amplifier tube (VT_1) divides at point 1; the middle and high frequency signals in the audio range take the path through the .001. mfd. condenser C_8 to audio amplifier tube VT_{3} , but the reactance of this condenser is so high at low frequencies that very few low frequency signals get through. The other path for signals from point 1 is through .5 mfd. condenser C_2 and through 100,000-ohm resistor R to audio amplifier tube VT_2 . Both low and high frequencies can pass through this condenser and resistor, but the high frequencies take

the .05 mfd. shunt path (C_4) around the .5 megohm potentiometer (R_3) to ground and consequently do not act upon the grid of VT_2 . Only the low frequency components and a portion of the medium frequencies are amplified by VT_2 . Resistor R prevents C_2 -R- C_4 from being a complete short for high frequencies, for that would leave none for the treble tube VT_3 . The potentiometers in the grid circuits of VT_2 and VT_3 can be adjusted to provide any desired relationship between the bass, medium and treble frequencies. The outputs of the bass and treble amplifier tubes are combined at point 2

accentuate the bass and treble frequencies, thus giving high-fidelity reproduction of music at normal low sound levels.

The circuit diagram of a typical Motorola tone control system is shown in Fig. 8. The tone control switch is shown at position V (VOICE). Under this condition the A.F. signal is developed across load resistor R_1 , all R.F. components accompanying this signal being by-passed to ground by the .0002 mfd. shunting condenser C_2 . This audio signal voltage is then applied to .5 megohm resistor R_2 through blocking condenser C_3 and by-pass conden-



FIG. 7. Special tone control circuit which uses separate channels for amplification of bass and treble frequencies, with independent controls for each channel.

and fed from there to the grid of audio amplifier tube VT_4 .

The Motorola Acoustinator. A rather unique bass and treble tone control is employed in some Motorola radio receivers. The audio systems in these receivers are designed to have a peak response at the medium frequencies, this being considered a highly desirable condition for clear voice reproduction. Means are provided for cutting down the medium frequencies and to some extent the treble frequencies when bass compensation is desired, such as at low levels of loudness. Means are also provided for cutting down the medium frequency response so as to

ser C_4 . Since R_4 and R_5 offer no paths to ground, the entire signal voltage across R_2 is applied to the control grid of the second A.F. tube through the R_4 - R_5 - C_5 combination. The output of this tube feeds into the primary of an audio transformer across which is connected .007 mfd. condenser C_9 . This condenser shunts some of the high-frequency audio signals around the primary, and therefore serves to cut down the high-frequency response. The audio amplifier normally has reduced bass response. As a result, when the tone control is in position V, the audio amplifier system has a response which is peaked in the middle frequencies.

Since the majority of voice frequencies are in this middle region, we have a condition which many people consider ideal for the reproduction of voice.

With the tone control in position M(MUSIC), the middle frequencies are cut down considerably and bass and treble frequencies therefore predominate. Position M is therefore used chiefly for reception of music at a low level of loudness. In this position of the tone control the bass signal currents take the path through resistor R_4 and condenser C_7 to ground, for C_5 is essentially an open circuit at low frequencies. The reactance of C_7 at bass reactance of C_5 is also quite low in relation to the ohmic values of R_4 and R_5 so at high frequencies we can consider R_4 and R_5 as being in parallel. The tone control input and output voltages will thus be the same at high frequencies, but this voltage will be reduced because R_4 and R_5 load the detector circuit. High and low-frequency voltages are about normal, while medium frequency voltages are cut considerably. The final effect is therefore a boosting of lows and highs.

With the tone control in position B, two condensers (C_6 and C_8) are connected to ground by the switch. C_6



FIG. 8. Heavy lines indicate the circuit of the Acoustinator tone control system as used on the Motorola Model 70 and on many other Motorola receivers.

frequencies (about 50 cycles) is approximately .6 megohm or six times the resistance of R_4 , and consequently nearly all of the bass signal voltage is developed across C_7 and is impressed upon the grid of the type 42 tube through R_5 . The middle frequencies, taking the same path as the bass signals, find a considerably lower reactance at the .0055 mfd. condenser; most of the available medium frequency signal voltage is wasted across R_4 , and only that small portion which is developed across C_7 affects the grid of the type 42 tube. At high frequencies, the reactance of C_7 is so low that it can be considered a short-circuit path. The acts in much the same way as C_7 in cutting down the response at medium frequencies, while C_8 acts like C_9 in cutting the highs. As a result, position B (BASS) gives a high bass response; this reduces static and interference noises to a minimum, and gives a soft mellowness to music.

The General Electric Tone Monitor. When an audio signal is fed from the output of an audio amplifier back into the input of an audio stage in such a way that the feed-back voltage is 180 degrees out of phase with the normal input signal for that stage, degeneration (reduction in output) will occur. In the tone control system shown in Fig. 9, used in General Electric model F-77 and other G.E. receivers, degeneration is purposely introduced in this way to provide a control over tone.

First let us analyze circuit conditions when the tone monitor switch (a type which connects together two adjacent contacts), is in the S or speech position. Observe that the switch now shorts R_{16} and C_{15} , so that feed-back current from point 1 on the secondary winding of the output transformer flows through the switch, through resistor R_6 to point 2 on the volume control, and then through section \mathcal{B} -G of the volume control to ground, developing across this section an out-of-phase voltage which is applied to the grid of the first A.F. stage through section \mathcal{Z} -P of this control. The higher the feedback current, the more degeneration or reduction in signal there will be. Since there is no reactance in the feed-back current path (only the 22,000-ohm resistance of R_0 , the amount of feedback current which causes degeneration will be constant over the entire frequency range. This will lower the gain over the entire frequency range, but will not change the shape of the frequency response curve for the amplifier.

The audio frequency response curve for the amplifier circuit in Fig. 9 is that represented by curve S in Fig. 10 (the heavy solid-line curve); this will also be the response curve when the tone monitor is in the S position, for the shape of the curve is not affected by uniform reduction of gain over the entire frequency range. Note that there is a peak in the high-frequency region from 3,000 to 6,000 cycles; this occurs because the primary leakage reactance of output transformer primary $T_{\rm s}$ acts with Condenser C_{20} to form a resonant circuit at these frequencies, and maximum power is fed to the input of T_{\star} . The peak response in this range, along with uniform response down to 100 cycles, provides a pleasing reproduction of speech at normal room volume.

When the tone monitor switch is placed at position F, the foreign-reception position, R_{16} and C_{15} still remain shorted. Condenser C_{19} is now in parallel with resistor R_6 ; at low frequencies the reactance of this condenser is quite high, with the result that feed-back current will take the path through R_6 . Normal amplifier response can therefore be expected at low frequencies, for feed-back current is



FIG. 9. The G. B. Tone Monitor Contrel eircuit is shown here in heavy lines, as applied to the Geaeral Electric Model F-77 receiver (the samplifier circuit is shown here in simplified form).

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still at the value determined by R_6 . At higher frequencies the reactance of C_{19} is naturally less; since this condenser shunts R_6 , more feed-back current flows at higher frequencies than would ordinarily be passed by R_6 alone, increasing the degeneration and thus cutting down the high frequency response. The shape of the response curve for this foreign-reception setting is represented by curve F in Fig. 10; this differs from the normal response curve only at frequencies above about 1,000 cycles. The attenuation or cutting of the high-frequency components serves to remove objectionable noise signals which ordinarily accompany

programs received from foreign countries.

When the tone monitor switch is in the N or normal-reception position, feed-back current flows through the parallel combination of R_{16} and C_{15} , and then flows through R_6 to point 2 on the volume control. At low frequencies the reactance of C_{15} is high, and R_{16} acts in series with R_6 to reduce the feed-back current practically to zero. As a result, peak bass response is secured. At medium and high frequencies the reactance of C_{15} becomes so low that it shorts out R_{16} , and only R_6 is effective in limiting feed-back current. This means that in the Nposition of the tone monitor switch we will have the normal treble peak in the response curve as well as a peak in the bass or low frequency region, as indicated by curve N in Fig. 10; this is a desirable condition for listening to music at the low loudness levels usually preferred in the home.

Setting the tone monitor switch at position B, the bass position, places condenser C_{19} across both R_6 and R_{16} , and inserts the R_{16} - C_{15} combination in series with R_6 . At low frequencies the reactances of these two condensers are so high that we can neglect them; this leaves R_{16} and R_6 acting in series and reducing feed-back current almost to zero. Degeneration is thus almost entirely eliminated at low frequencies, and bass response goes up.

At medium frequencies the reactance of C_{15} is low enough to provide a shunt path around R_{16} , but C_{19} is still out of the picture. The 22,000ohm resistance of R_6 now controls feedback current, and so we have essentially the same medium frequency response as was secured at positions S, F and N.

At high frequencies C_{15} shorts out R_{16} , and C_{19} provides a low-reactance

shunt path around R_6 . Feed-back current thus increases considerably, cutting down the high-frequency response quite rapidly as frequency increases. At position B, then, curve B in Fig. 10 will be the response curve of the receiver; it clearly indicates a boosting of bass frequencies.

Naturally each setting of the tone monitor control will sound different to the human ear at each setting of the volume control, because of the changing characteristics of the human ear with loudness. Automatic compensation for the variations in sensitivity of the ear with loudness is secured by



FIG. 10. This graph shows how the G. E. Tone Monitor Control (Fig. 9) changes the sudio frequency response of the receiver in which it is used. The graph represents full-volume conditions, when P is at point 3; there is then no sutomatic tone compensation.

making the tone control connection to a tap at point 2 on the volume control potentiometer. Varying the volume control setting thus gives an automatic balance between the normal response of the amplifier and the special response introduced by the tone monitor circuit.

Volume Expansion Circuits

The compression of sound level is of extreme importance in the practical operation of radio transmitters, for otherwise over-modulation and an excessively low signal-to-noise ratio would exist. Compression of sound level is just as important in the production of disc recordings of musical numbers, for excessive volume results in over-cutting of the grooves. Volume compression, of course, destroys the fidelity or faithfulness of a program, for it cuts down the volume of highlevel passages. In order to counteract this volume compression, it is necessary to introduce some form of volume expansion in the reproducing device.

Although automatic volume expansion can be applied to both radio receivers and electric phonographs, it is far more satisfactory when the original compression is performed automatically, as is now being done on most recordings. In radio transmitters the compression is done manually by the studio operator, and obviously no automatic circuit in the receiver can exactly compensate for an action which depends for its accuracy upon human alertness and skill.

The most widely used system for automatic volume expansion depends upon the fact that the gain of a variable mu or super-control pentode tube varies with the negative C bias voltage. which is applied. By applying to a variable mu pentode amplifier tube, in series with its normal negative C bias, a positive D.C. voltage which increases with sound level, the gain of the tube can be made to increase as sound level or volume increases. The necessary D.C. or A.V.E. (automatic volume expansion) control voltage is secured by rectifying the audio voltage and filtering out the A.C. component of the resulting pulsating voltage. This D.C. voltage must of course be time-delayed, for if the A.V.E. control voltage varies too rapidly with fluctuations in sound level, the reproduced music will sound "choppy" and have a "gurgling" effect; likewise if the A.V.E. control voltage is too greatly time-delayed, the benefit of automatic volume expansion

will be lost.

RCA A.V.E. Circuit. A typical automatic volume expansion circuit based upon this principle and used in a phonograph amplifier is shown in Fig. 11. The phonograph pick-up at the upper left in the diagram feeds an audio signal into step-up transformer T_1 , across the secondary of which is shunted a 625,000-ohm volume control potentiometer R_4 . Notice that this potentiometer is in an automatic tone compensating circuit, there being two taps on the potentiometer for this purpose. Series resonant circuit L_2 - C_2 , connected to the first or uppermost tap, serves to attenuate the middle frequencies and thereby has the effect of raising the bass and treble response at medium and low volume levels. Resistor R_2 broadens the resonant response of L_2 - C_2 and limits the amount of attenuation at medium frequencies. R_3 and C_3 , connected between the lower tap on the potentiometer and ground, provide an additional boosting effect for the bass response at still lower volume levels. Series resonant circuit L_1 - C_1 , also connected across the secondary of T_1 , resonates at about 700 cycles and serves to cut down the response of the audio amplifier in this region. Resistor R_1 prevents complete cut-off at resonance.

After passing through the volume and tone control circuits, the audio signal is fed through D.C. blocking condenser C_4 into vacuum tube VT_1 , which is ordinarily operated at a high negative C bias and thus normally has low gain. At the same time the audio signal developed across the secondary of transformer T_1 is fed into vacuum tube VT_2 , a triode amplifier; this audio signal has not been acted upon by the automatic tone and volume control circuits. The output of VT_2 is rectified by VT_3 and then filtered to give a D.C. voltage whose value is dependent only upon the signal level at the secondary of T_1 . This varying D.C. voltage is applied to one of the grids of pentode tube VT_1 in such a way as to increase the gain of the tube in proportion to the level of the original sound. Now let us trace the circuit in greater detail.

Potentiometer R_8 controls the amount of signal voltage applied to triode amplifier VT_2 , and thus provides a control over the amount of volume Since VT_3 permits electron flow only in the direction from cathode to grid, the A.F. voltage existing across R_{10} sends through R_{11} a pulsating current which develops across this .22-megohm resistor a pulsating D.C. voltage, with polarity as indicated. Condenser C_{12} smooths out the A.F. variations, and the resulting D.C. voltage is applied to the third grid (G_3) of VT_1 through filter R_{13} - C_5 , which provides the necessary time delay in the application of



FIG. 11. Automatic volume expander circuit as used in a number of RCA electric phonographs.

expansion. R_9 and C_9 together provide automatic C bias for this tube, and plate voltage is obtained by a connection through resistor R_{12} to the B_+ terminal of the receiver power supply circuit. The A.C. plate current of VT_2 flows through C_{10} , R_{10} and C_{11} to the chassis, with most of the voltage being dropped across R_{10} . Observe that the plate and cathode of VT_3 are connected together; this tube therefore acts as a diode rectifier with its grid (acting as anode) connected to one side of R_{10} and its cathode connected through R_{11} to the other side of R_{10} . the A.V.E. control voltage. The negative terminal of R_{11} connects to ground through potentiometer R_{15} , which is connected across a part of voltage divider R_{14} in such a way as to provide an adjustable negative C bias which will act in series with the positive bias across R_{11} . Potentiometer R_{15} need be adjusted only when a new tube is inserted in the volume expander circuit.

At low volume levels practically no D.C. voltage is developed across R_{11} , and grid G_8 of VT_1 then receives the full negative C bias voltage developed by R_{15} . As the volume level increases,

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the positive voltage drop across R_{11} increases and counteracts the negative C bias, with the result that the third grid becomes less negative and the gain of tube VT_1 increases.

The action of an automatic volume expander circuit of the type shown in Fig. 11 is illustrated by the graph in Fig. 12. The lower curve tells how the output level varies with the input level when there is no automatic volume expansion, and the upper curve gives the



FIG. 12. Automatic volume expansion action of the phonograph amplifier circuit in Fig. 11. In proparing this graph, a definite low input signal level was assumed and called 0 db input level, and the output level under this condition was called the 0 db output level. These curves tell that if the receiver is being fed with a signal of 35 db (point A), the output level without A.V.E. will also be 35 db (point B); with A.V.E., however, the output level will be 20 db higher, or 55 db (point C).

same information for a circuit having automatic volume expansion. Notice that as the input signal level is increased, automatic volume expansion makes the output level increasingly greater than would be obtained without A.V.E. With proper circuit design, an exact reproduction of the original program can be secured even though the volume level is condensed during the recording process in the case of a phonograph record or during the transmitting and receiving process in the case of a radio system.

Crosley Auto-Expressionator. Another interesting automatic volume expansion circuit, which in addition provides automatic bass compensation at low sound levels, is that shown in Fig. 13; as you can see, this is a special form of filter (some engineers call it a differential bridge) connected between the output transformer of the receiver and the loudspeaker. It is used in a number of Crosley receivers, where it is known as an *auto-expressionator*.

All four switches in Fig. 13 are operated by a single control which permits instant change-over from normal operation to A.V.E. and A.B.C. Normal operation is secured when switches SW_3 and SW_4 are open, and switches SW_1 and SW_2 are closed; a study of the circuit will show that under these conditions a direct connection exists between the loudspeaker and the output transformer, with no parts in series or shunted across the loudspeaker.

With the control switch in the A.V.E. position, switches SW_1 and SW_2 will be open and switches SW_{3} and SW_{4} will be closed, as indicated on the circuit diagram. We will assume for the moment that the loudspeaker is disconnected. Audio frequency current flowing from the output transformer to point A will have two paths to point B; 1, through X_1 (a ballast resistor), resonant circuit L_1 - C_1 and R_1 ; 2, through R_2 , resonant circuit L_2 - C_2 and ballast resistor X_2 . At point B the currents from the two paths combine again and return to the output transformer. Resistors R_1 and R_2 are equal in value; ballast resistors X_1 and X_2 , whose resistance increase with current, are likewise the same size; resonant circuits L_1 - C_1 and L_2 - C_2 are also identical in characteristics, and thus the same current would flow over each path when the loudspeaker is disconnected.

When the loudspeaker is connected between points C and D, the loudspeaker current will depend upon the voltage existing between these points, and this voltage will be the difference between the voltage drops for A-C and the voltage drop for A-D. The ballast resistors used in this circuit are of special design, increasing in ohmic value as the current through them increases. This means that as the output current of the receiver increases, the resistances of X_1 and X_2 increase, causing the impedance of branch A-C to be greater than that of branch A-D and causing the impedance of branch B-D to be greater than that of B-C. Even when cold, the ohmic values of the ballast resistors are slightly greater than the ohmic values of R_1 and R_2 . As a result, the circuit is in an unbalanced condition at all times, with some signal voltage always being applied to the loudspeaker. The unbalance becomes greater as the receiver output current increases and sends more current through the ballast resistors. At low values of power output, when the circuit is most nearly balanced, only a small amount of the total output power is supplied to the loudspeaker.

As the power output of the receiver increases, the circuit becomes farther out of balance and more and more of the receiver output power is applied to the loudspeaker. This increases the intensity of the louder passages, expanding the volume range. The time lag in the heating and cooling of each ballast resistor filament is an important factor in the proper operation of this circuit, for it determines the speed of A.V.E. action.

Now let us see how this circuit works as a bass compensator. Since each resonant circuit resonates at about 40 cycles, we can see immediately that for the medium and high audio fre-

quencies, the reactances of condensers C_1 and C_2 become quite low, and the resonant circuits thus do not create an unbalance at these higher frequencies. At low or bass frequencies, however, the resonant resistance of each resonant circuit becomes quite high, causing a greater-than-normal unbalance and thus sending more current through the loudspeaker. The result is a boosting of the bass notes at low sound levels. At high sound levels, however, the resistance of X_1 increases to the point where it makes the resonant resistance of each resonant circuit prac-



FIG. 13. Crosley Auto-Expressionator circuit, a combination automatic volume expansion and automatic bass compensation arrangement.

tically negligible in so far as unbalance is concerned; this means that bass compensation occurs only at low sound levels.

The Noise Problem

Signal-to-Noise Ratio. One of the most perplexing problems with which the radio engineer has to deal is that of noise. Regardless of whether this noise originates in the receiver or outside the receiver, it is reproduced along with the desired signal and is oftentimes strong enough to be annoying. Recognizing that it is impossible to eliminate noise entirely, the engineer endeavors to use apparatus which will make it negligible with respect to the **desired** code, sound or picture signal. This is why he continually refers to the signal-to-noise ratio (the signal voltage or current divided by the noise signal voltage or current).

Even the average radio listener unknowingly recognizes the importanceof a high signal-to-noise ratio, for he invariably tunes to a local station when he desires to enjoy a high-quality musical program with a minimum of noise. This local station, being close to the receiving location and having reasonably high power, can produce at the receiving antenna a high enough signal strength to give the desired high signal-to-noise ratio.

Man-Made Interference Noises. Noise originating outside of the receiver is usually caused by electrical apparatus such as motors, sign flashers and other devices in which sparks occur when electrical circuits are broken. This type of noise is commonly referred to as man-made interference and is most common in industrial cities and towns, where it will predominate over atmospheric noises except perhaps on stormy days. Much of this interference can be reduced by the use of special noise-reducing antennas (considered elsewhere in the Course), but the ideal remedy is elimination of the interference at its source by the use of noise filters.

Radio stations have aided materially in reducing the effects of noise by increasing the power of their transmitters. In extremely noisy locations in cities, owners of radio receivers should be told to tune to the high-power local stations in order to secure a high signal-to-noise ratio at their receivers.

The wave form of a typical noise signal is shown in Fig. 14; notice that there is a general noise level, existing practically all of the time, which is ordinarily not objectionable except when low-power or distant stations are

tuned in; in addition, there are sharp pulses of noise or static at irregular intervals, which usually are objectionable. As a rule these pulses are of very short duration, less than .001 second. -Receiver Noise -- Inside -the-radio receiver, thermal agitation and tube shot effect, particularly in the first R.F. stages of a receiver, result in unavoidable circuit noises. Fortunately these circuit noises can be reduced to a level less than that produced by a onemicrovolt input signal (by proper selection of circuits and tubes), and the signals of most stations can override this noise level. The mixer-first detector in a superheterodyne receiver can give excessive noise, but a properly



FIG. 14. Wave forms of typical rectified noise signals as they may exist in the detector stage of a radio receiver.

operated stage preceded by R.F. amplification will not have this trouble.

We shall now consider in turn the use of tone control circuits, noise impulse-silencing circuits and inter-carrier noise suppression circuits for noise-reducing purposes.

Noise-Reducing Tone Controls

An analysis of noise signals will reveal that they contain practically all audio signal frequencies, with the higher frequencies, above 3,000 cycles, predominating. It is for this reason that a tone control which suppresses or attenuates the high audio frequencies is also effective in reducing the amount of noise. Of course, this suppression of the high audio frequencies destroys the faithfulness of reproduction, and consequently tone controls are far from being a satisfactory solution to the noise problem.

Noise Impulse-Silencing Circuits

The Lamb Noise Silencer. The circuit shown in Fig. 15, developed by J. J. Lamb, serves to eliminate noise pulses by silencing the receiver for the duration of each pulse. This circuit has generally inserted between the I.F. amplifier and the second detector of a superheterodyne receiver, as indicated in Fig. 15.

The signal voltage developed by the second I.F. transformer between point A and ground is made up of an I.F. carrier modulated by noise signals and by desired intelligence signals. This voltage is applied directly to the con-



FIG. 15. The Lamb noise silencer, which silences the receiver for the duration of each strong noise pulse, is shown here incorporated in a superheterodyne receiver.

been found quite valuable in commercial and communication receivers which are used for code signal reception. It has not been entirely satisfactory for broadcast receivers, however, for the action of the noise-silencing circuit has the effect of partially destroying the peaks of modulation in the received carrier, thereby creating the more serious problem of amplitude distortion.

The Lamb noise-silencing circuit is

trol grid of the 6J7 noise amplifier tube, and also to the control grid of the 6L7 second I.F. amplifier tube. The amplified output of the second I.F. tube feeds through tuned primary transformer T_3 to the 6H6 full-wave diode second detector, which through rectification provides the desired A.F. signal voltage and a D.C. voltage for A.V.C. purposes. The amplified output of the noise amplifier tube, on the other hand, feeds through tuned primary
transformer T_2 (tuned to the I.F. value) to another 6H6 double-diode tube which serves as a full-wave noise rectifier. The cathodes of both the noise amplifier and noise rectifier tubes are connected to the movable contact of noise threshold adjustment potentiometer R_{12} , and the setting of this control determines the value of the negative bias which is applied to the control grid of the noise amplifier and to the diode sections of the noise rectifier. This bias on the diodes acts much like a delay voltage, in that it determines the minimum signal strength required to swing the diode plates positive and start rectification; the bias is normally set so that rectification occurs only for noise peaks extending above the desired signal level. Note that the noise amplifier tube also receives a bias from the A.V.C. circuit of the receiver through A.V.C. filter R_3 - C_3 ; this serves to maintain the proper noise threshold level despite normal variations in desired signal level due to fading.

The rectified noise voltage, which is developed across noise rectifier load resistor R_8 when strong noise peaks are present, is made up of I.F., A.F. and D.C. components. Filter combination C_{14} - L_7 - C_{12} removes the I.F. and A.F. components; the remaining D.C. voltage is applied to the third grid of the second I.F. amplifier-noise silencer tube, driving this grid more negative with respect to ground and thus blocking plate current flow for the duration of each strong noise pulse.

When no noise pulses are present, the noise rectifier plates are negative with respect to their cathodes; no current passes, no D.C. voltage exists across R_8 , and consequently the third grid of the 6L7 second I.F. tube has only the normal negative C bias determined by R_5 and R_6 . Desired signals are therefore amplified and detected in a normal manner by the two lower tubes in Fig. 15.

When a strong noise pulse enters the receiver along with the desired signals, it instantly drives the noise rectifier plates positive, and the D.C. voltage developed across R_8 by the resulting rectified current flow is applied without delay to the third grid of the second I.F. amplifier-noise silencer tube, blocking plate current flow completely. Under this condition no desired or noise signals whatsoever reach the second detector, and consequently the noise is not heard. The instant the noise pulse drops below normal signal level, the noise rectifier stops conducting, the blocking voltage is removed from the third grid of the 6L7 tube, and normal receiver operation is restored.

Naturally there must be no time delay between the arrival of a noise pulse at the noise amplifier and the application of the D.C. silencing voltage to the third grid of the noise-silencer tube if the silencing action is to be instantaneous and is to last only for the duration of each strong noise pulse. Under this condition the receiver is silenced for such a short interval at a time that the action can scarcely be noticed by the human ear.

Diode Noise Limiter. If the load of a diode detector is instantly shunted with a large condenser when a noise pulse comes through, the audio voltage developed across this load, as well as the noise pulse, will be greatly reduced for the duration of the pulse. A practical noise-limiting circuit based upon this principle is shown in Fig. 16. This circuit uses a type 6H6 double-diode tube, with section D_1 serving as second detector (resistors R_2 and R_3 form its load) and section D_2 serving as an automatic noiselimiter switch places a large condenser

across R_{a} , a part of the detector load, during the time of a noise pulse (the two sections are shown as separate tubes to simplify your study of the circuit, but remember that they are actually in the same vacuum tube envelope). The D.C. voltage developed across the detector load resistors is used for A.V.C. purposes, with C_F and $R_{\rm F}$ serving as the A.V.C. filter. Condenser C is an R.F. by-pass condenser which keeps I.F. signals out of the detector load. Switch SW is a panel control, and must be in the closed position when noise-limiting operation is desired.

When no strong noise pulses are present, the normal rectified electron flow is from point 1 to point 3 through the detector load, and point 1 is therefore more negative than points 2 and 3. The D.C. voltage developed between points 1 and 3 charges condenser C_1 through resistor R_1 , which is of high ohmic value. Once this condenser is charged, no additional current flows through R_1 , and points 1 and 4 are then at the same negative potential with respect to ground (point 3). Diode section D_2 now acts as a high resistance, . for its anode (point 4) is more negative than its cathode (point 2).

Now suppose that a strong noise pulse is rectified by diode section D_1 along with the desired modulated car-The flow of rectified current rier. through the detector load increases instantly, and the D.C. voltage between points 1 and 3 will likewise increase instantly. C_1 will begin drawing current through R_1 in order to become charged to the new voltage value. The time constant of the R_1 - C_1 combination is quite high, however (ordinarily it is from 1/10 to 1 second), and consequently C_1 cannot be charged instantly to the new voltage. Since the average noise pulse lasts about 1/1000 of a second, which is considerably less than the time constant, point 4 remains essentially at the previous low negative potential for the duration of the noise pulse, while point 2 goes instantly to the new higher negative potential. This means that the plate of diode section D_2 becomes positive with respect to its cathode for the duration of the noise pulse; current then flows through D_2 and its resistance drops to a quite low value. Since this tube is in series with high-capacity condenser C_1 across R_3 , the net impedance between points 2



FIG. 16. Diode noise-limiter circuit, which shunts all audio signals to ground through a condenser for the duration of each strong noise pulse when switch SW is in the closed or noise-reducing position.

and S is lowered, and the A.F. voltage across R_3 is therefore lowered. As a result, the A.F. voltage fed to the first audio stage drops considerably for the duration of each noise pulse, and the A.F. output voltage of the receiver is correspondingly lowered. The instant the noise drops below normal signal level again, point 2 returns to its original potential, current flow through D_2 stops, and the D_2 - C_1 shunt path across R_3 is no longer effective in limiting the audio amplifier input voltage.

Inter-Carrier Noise Suppression Circuits

In receivers having A.V.C., the noise which is heard when tuning between stations may be objectionable to some listeners. This inter-station or intercarrier noise occurs because automatic volume control raises the gain of the R.F. system to the point where normal external and internal noises become -objectionable. Some receivers have an inter-station noise-suppression circuit which works in conjunction with A.V.C. to silence the receiver when no station is tuned in. This inter-carrier noise suppression system is often known as quiet automatic volume control, abbreviated Q.A.V.C.; since it stops or squelches noise it is also known as a noise-squelching circuit. In some circuits the detector action is delayed for weak signals; in other circuits the I.F. or A.F. amplifier is blocked until the station signal reaches a level which is high enough to override noise which may be present, thus preventing reception of stations having less than the minimum satisfactory signal-to-noise ratio. To be sure, this has the effect of reducing the apparent sensitivity of the receiver; a switch is therefore provided to cut out this noise-suppressing action when maximum sensitivity is required regardless of noise.

The diode noise-limiter circuit shown in Fig. 16 also serves to a certain extent as an inter-carrier noise suppressor. Suppose that the receiver is tuned off a station. Diode section D_1 of course receives no input voltage under this condition, and no current flows through R_2 and R_3 . Points 1 and 2 immediately assume ground potential, and point 4 will likewise be at ground potential after the time-delay interval (less than one second), during which condenser C_1 discharges to ground through R_1 , R_2 and R_8 , has elapsed.

Noise or static signals which enter the receiver under this condition will, of course, be rectified by diode section D_1 , making point 2 negative with respect to ground. Point 4 remains at ground potential because of time-delay action, making the plate of D_2 positive with respect to its cathode. Noise signals reaching point 1 will therefore take the low-impedance path through R_2 , D_2 and C_1 to ground. Noise is thus diverted from the input of the A.F. amplifier and will not be reproduced by the loudspeaker when tuning slowly from station to station. Of course, the noise-reducing action of this circuit ceases as soon as the time-delay interval has elapsed.

In the more conventional types of inter-carrier noise suppression circuits, the receiver is made inoperative for weak signals on the assumption that the noise signals heard between stations will be weaker than any of the desired strong carrier signals. Three types of inter-carrier noise suppression circuits are used for this purpose: 1, biased demodulator circuits, which place a bias or delay voltage on the diode detector so that only carrier signals above a certain level will be demodulated; 2, A.F. amplifier blocking circuits, which block or prevent operation of the A.F. amplifier until the carrier signal exceeds a definite minimum value; 3, I.F. and R.F. amplifier blocking circuits, which prevent operation of these amplifiers until the carrier level becomes appreciably higher than the noise level. Let us consider one example of each of these inter-carrier noise suppression methods.

Biased Demodulator Circuit. As you know, delayed A.V.C. action can be obtained by inserting in the separate A.V.C. diode rectifier circuit a delay voltage which must be exceeded by the carrier signal before the rectification required for the production of an A.V.C. voltage can take place. Usually, a separate diode rectifier is then required for demodulation purposes.

If we introduce this delay voltage directly into the diode rectifier circuit used for demodulation, so that only signals above a certain minimum level will be rectified, we have one form of inter-carrier noise suppression. Weak carrier and noise signals cannot overcome this delay voltage, and hence are not reproduced by the loudspeaker as the receiver is tuned from one station, to another. With this arrangement, however, the receiver must be tuned to fairly strong carriers in order to secure satisfactory reproduction, for mediumstrength signals will undergo a certain amount of amplitude distortion.

A circuit incorporating this biased demodulator type of inter-carrier noise suppression is shown in Fig. 17. In the D (distance) position of switch SW, the two diode sections of the 6B7 tube together serve for demodulation and A.V.C. purposes, and there is no delay voltage in the diode circuit. Under this condition all signals, including noise, are detected in the usual manner.

When the switch is placed in the L(local) position, the voltage drop across cathode bias resistor $R_{\rm C}$ acts in series with the diode circuit. This delay voltage must be exceeded by the signal before A.V.C. action and demodulation can take place, and the carrier signal must be reasonably larger than this delay voltage before complete demodulation can occur. If $R_{\rm C}$ (which acts also as C bias resistor for the amplifier section of the 6B7 tube) is replaced with a potentiometer and the movable arm is connected to the L terminal of the switch, the amount of delay voltage in the rectifier circuit can be varied.

A.F. Amplifier Blocking Circuit. In

Fig. 18 is a well-known example of a circuit which employs quiet automatic volume control (Q.A.V.C.) action to block or cut off the input signal to the A.F. amplifier whenever the carrier level becomes so low that noise will be objectionable. As you can see, a special Q.A.V.C. tube is inserted between the detector and the first A.F. stage, with a switch in the cathode lead to make this tube inoperative when ordinary receiver action is desired.

Observe that the plate-cathode resistance of the Q.A.V.C. tube acts in series with a 1-megohm resistor (part



FIG. 17. Biased demodulator type of inter-carrier noise suppression circuit, as used in the Silvertone Model 7149 receiver. This circuit reduces the noise ordinarily heard in an A.V.C.-controlled receiver when tuning between strong stations.

72) between B+ and ground G, with the screen grid of the type 77 A.F. amplifier tube connected to point X as indicated. We thus have a voltage divider, with the screen grid voltage of the type 77 tube being determined by the plate-to-cathode resistance of the Q.A.V.C. tube. The lower this resistance, the lower will be the positive voltage developed across the Q.A.V.C. tube and applied to the screen grid of the A.F. tube, and the lower will be the gain of this type 77 tube.

Now let us see how weak signals lower this plate-cathode resistance and

thereby block the first A.F. tube so as to suppress inter-carrier noises. Notice that the control grid of the Q.A.V.C. tube connects to the diode second detector load through a 4-megohm resistor. Under a no-carrier-signal condition, the only D.C. voltage across diode load resistor 55 will be a very low D.C. voltage produced by noise signals; condenser 52 and resistor 53 remove all A.C. component which might otherwise reach the Q.A.V.C. tube. The Q.A.V.C. tube thus gets a small negative C bias, under which condition its plate current is high and its plate-to-cathode resistance quite low. This makes the screen

this tube and thereby increasing the screen grid voltage on the 77 tube enough to permit amplification of the desired signals in the usual manner.

In the circuit just described, blocking action was secured by varying the screen grid voltage of the first A.F. amplifier tube; the same results can be secured by applying the blocking voltage to the control grid of this tube, as is done in the inter-carrier noise suppression circuit in Fig. 19. No extra tube is required here, for the triode section of double-diode-triode tube VT_1 is used to produce the required blocking voltage. A desired carrier signal



FIG. 18. This quiet A.V.C. circuit, used in the Philco Model 16-122 receiver, acts upon the screen grid of the first A.F. tube, blocking the amplifier whenever the receiver is tuned between strong stations. It thus serves to suppress inter-carrier moise.

grid voltage of the type 77 tube low enough to prevent amplification of any noise signals which are fed to the control grid of this tube through D.C. blocking condenser 57. Potentiometer 82 controls the screen grid voltage on the Q.A.V.C. tube and thus determines the amount of plate current which will flow during this low negative C bias condition. This potentiometer can be adjusted so that signals below any desired level will be blocked.

When a desired strong carrier signal reaches the detector, the resulting D.C. voltage across detector load resistor 55increases the negative bias on the control grid of the Q.A.V.C. tube, increasing the plate-to-cathode resistance of is rectified by the combined diode sections of the tube in the usual manner, developing a D.C. voltage across detector load resistor R_2 . The audio voltage which also exists across this resistor is fed through D.C. blocking condenser C_5 to the control grid of the pentode first A.F. amplifier tube VT_2 , while the D.C. voltage is fed through resistor R_3 to the control grid of the triode section of tube VT_1 . Plate current for this triode section flows through resistor R_4 , producing across it a voltage drop which acts with the drop across voltage divider resistor R_{9} to provide C bias for VT_2 . You can readily see that when the current through R_4 is small, the C bias on the

pentode tube will essentially equal the voltage drop across R_9 which is ordinarily about 2 volts. This is the C bias required for normal A.F. amplification of a desired signal.

When no carrier signal is being rectified by the diode section, the D.C. voltage across R_2 is essentially zero, which means that the triode section of VT_1 will be operating at zero bias. This makes the plate current of this section high, and this large current flowing through R_4 increases the negative C bias considerably on the control grid of VT_2 . As a result, the pentode canA.F. amplifier blocking circuit. In Fig. 20 is shown a circuit which, when switch SW is closed, will accomplish the desired results, driving all of the control grids in the R.F. and I.F. sections highly negative for weak signals and placing a negative bias on the second detector diode sections to prevent rectification of weak signals. Normal bias is restored immediately when the carrier signal becomes greater than the minimum desired level.

An analysis of this circuit starts with resistor R_{13} , across which the second detector develops the desired audio



not amplify any weak signals or noise signals which may appear across R_2 as the receiver is tuned between stations.

When a desired strong carrier signal is tuned in, the D.C. voltage developed across R_2 drives the grid of the triode highly negative, reducing the plate current of VT_1 and restoring the C bias of VT_2 to its normal value.

I.F. and R.F. Amplifier Blocking Circuit. Noise signals and weak carrier signals which are ordinarily heard when tuning between stations can be suppressed by applying to the I.F. and R.F. sections the same principle as that just described in connection with the voltage and the D.C. voltage for A.V.C. purposes; the plate current of the whisper control tube also flows through R_{13} , developing across it an additional voltage for blocking purposes. When no carrier signal is present, the only voltage across R_{13} is that due to plate current of the whisper control tube, and the magnitude of this voltage is determined by the setting of movable arm P on potentiometer R_{18} and by the negative bias on the control grid of this tube. Point 1 will be negative with respect to the chassis during this nocarrier-signal condition.

Now observe that the output of the I.F. amplifier is fed to the upper diode

of the whisper control tube through C_{19} . When no carrier signal is present, there will be no rectified current flow, and point X and the control grid of the whisper control tube will therefore be at cathode potential; as a result a large plate current flows through the triode section of this tube, then through SW, R_{13} , R_{16} , R_{17} , R_{18} up to P and back to the cathode. This develops a large negative voltage across R_{13} and since this voltage is applied to the A.V.C.-controlled tubes through the usual A.V.C. feed system, it blocks these tubes.

When weak intelligence and noise signals are present in the output of the I.F. amplifier, such as when tuning between powerful local stations, these pass through C_{19} to the upper diode section of the whisper control tube, where they undergo rectification and produce D.C. voltage drops across resistors R_4 , R_{29} and R_7 . The direction of electron flow is such that point Xon R_7 will be negative with respect to the cathode of the tube; since the control grid of the triode section of the whisper control tube is also connected to this point, it gets a negative C bias. Plate current flow is thereby reduced, and the voltage drop produced by this plate current is correspondingly reduced. This reduces the negative bias applied to the R.F. and I.F. sections through the A.V.C. system, but the negative bias on the second detector diode sections is still high enough to prevent signal rectification.

When a sufficiently strong carrier signal is tuned in, the diode sections of the whisper control tube will pass enough current to block plate current flow through the triode section (by increasing the voltage drop across R_7), and practically no current will be drawn through R_{13} by the whisper control tube. This completely releases the delay action of the whisper control tube upon the A.V.C. circuit, restoring normal high sensitivity of the receiver. A.V.C. action remains normal with further increases in carrier level. The setting of potentiometer R_{18} determines the minimum carrier level at which the delay action is released and normal A.V.C. action begins.



FIG. 20. This inter-carrier noise suppression circuit used in the Silvertone Model 4666 receiver where i is called a "whisper tuning control," serves to block the R.F. and I.F. amplifiers through the A.V.C. system, and at the same time provides delayed A.V.C. action.

The circuit in Fig. 20 thus provides a combination of inter-carrier noise suppression and delayed A.V.C. action. The listener is able to tune from one station to another even at low sound levels or whisper levels without hearing inter-station noise, and thus this circuit has been called a whisper control circuit. Opening switch SW removes the delay action, allowing reception of weak signals along with noise. Observe that the opening of this switch does not stop the functioning of the diode section of the whisper control tube; this is essential because the diode also provides a D.C. control grid voltage for the electron ray indicator tube or tuning eye.

Lesson Questions

Be sure to number your Answer Sheet 24FR-1.

Place your Student Number on every Answer Sheet.

Never hold up one set of lesson answers until you have another ready to send in. Send each lesson in by itself before you start on the next lesson. In this way we will be able to work together much more closely, you'll get more out of your Course, and you will receive the best possible lesson service.

- 1. What three types of circuits are used for reducing the annoying effects of atmospheric noises which enter a radio receiver?
- Ine Costrol Circuite, None inpulse silencing Circuite. Super man
 2. At low loudness levels, is the human ear more sensitive to medium-frequency fully sounds (500 cycles to 5,000 cycles) than to low-frequency sounds (below 500 cycles)? (44 500 to 500 c)
- 3. Why does a simple condenser tone control give an apparent boost in bass the signals? LOW all the control provide trable control of the sector of
- 4. How can a conventional tone control provide treble control? "I're ~
- 5. What type of circuit is present when the volume control has a fixed tap which is connected to ground through a condenser and a resistor?
- 6. Why is degeneration intentionally introduced in the General Electric Model F-77 receiver circuit shown in Fig. 9. To provide a Control
- Upon what fact does the operation of the most widely used automatic volume expansion system depend? p-/6
- 8. Why should a receiver be tuned to a local station when a minimum of noise is desired? Because you get the discuss high
- 9. Why is inter-carrier noise heard in receivers having A.V.C.? P-L4
- 10. Name the three conventional types of inter-carrier noise suppression circuits.

HONESTY

That old proverb, "Honesty is the best policy," is just as true today as ever. Any firm which depends upon repeat business cannot exist for long without following this policy; any man who deals with other people cannot afford to disregard it.

Strict observance of the law will keep a man out of jail, but that does not necessarily make him an honest man. Honesty goes far beyond the law; it involves a careful regard for the rights of others, a truthfulness and sincerity in dealing with others, and a fairness and trustworthiness in matters involving property or business.

It is not enough to act so others will think you are honest; you yourself must know that you are playing the game fair and square if you are to enjoy that real satisfaction associated with absolute honesty.

Be honest, and your reputation will take care of itself. Let your spoken word, your slightest implied action be as good as your signature on a legal contract, and you will enjoy those things which no amount of money can buy—happiness, success, and the respect of your fellow men.

J.E. SMITH



Study Schedule No. 25

For each study step, read the assigned pages first at your usual speed. Reread slowly one or more times. Finish with one quick reading to fix the important facts firmly in your mind, then answer the Lesson Questions for that step. Study each other step in this same way.

Here you learn about the equipment which makes possible almost magical feats in electronics. The apparatus used, the fundamentals of light and illumination, and the types of photoelectric cells arc all described. Answer Lesson Questions 1, 2 and 3.

There are two types of these cells—the vacuum and the gas-filled types. The small amount of gas in the latter type changes the characteristics greatly. Answer Lesson Questions 4 and 5.

- 4. Photovoltaic CellsPages 18-24

This cell generates its own voltage, so does not require a power source for its operation. This is very important and leads to many applications of these cells. They are used particularly in exposure meters and light meters. Answer Lesson Questions 6, 7 and 8.

- **5.** Electron-Multiplier Cells Pages 24-28 The extremely high sensitivity of these cells makes them very useful where small light values are encountered. They do require a high-voltage power supply, but this is outweighed in many applications. Answer Lesson Questions 9 and 10.
 - 6. Mail your Answers for this Lesson to N.R.I. for Grading.

7. Photoelectric Control Circuits with Relays. Reference Text 25X

Here's a fascinating subject which is becoming of increasingly greater importance as the industrial world turns to electronics for its automatic controls. Read it once, to familiarize yourself with the contents of the book. You will then know where to find this information when you actually need it. If you're doing electronic work now or are deeply interested in it, however, you'll want to study this book just as thoroughly as a regular lesson.

8. Start Studying the Next Lesson.

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Ρ.

LIGHT-SENSITIVE CELLS FOR CONTROL CIRCUITS

The Electric Eye

NDOUBTEDLY you have read about the magic electric eye, a seemingly mysterious device which causes doors to open as you walk toward them, turns on roadside signs in the country as an automobile approaches. sounds alarms when anyone walks over a forbidden area, prevents dangerous machines such as punch presses from operating while the hands of the worker are in the way, and does thousands of other equally amazing and practical feats. In this lesson you will learn about the many different types of electric eyes, which are called photoelectric or light-sensitive cells by technical men. You will learn how each type of cell is constructed, how it functions, and how it can be made to replace man's eyes.

► Thousands of light-sensitive cells are in use today in every corner of the world, responding to beams of light which may be perfectly invisible to the human eye, detecting every change in illumination from the sun or from other sources of light: these cells change their *electrical* characteristics with variations in the light which they "see". These light-sensitive cells start and stop heavy machinery, count objects moving past at a mile-a-minute speed, guard against fire, smoke, water, and burglars, and even "read" books for blind persons. Cigars, beans, eggs, fruit, and other products are being graded as to color or shade by light-sensitive cells, faster and more

accurately than by the human eye.

Although the field of photoelectricity is not new, its development into commercial practicality has taken place within the last few years. Scientists have known for more than one hundred years that certain electrical effects could be obtained by exposing chemical elements and com-



Courtesy General Electric Co.

A typical industrial use of photoelectric equipment. When the light beam is interrupted by a box, a number of actions are possible. The unit may actuate a counter; may sound an alarm; or may operate automatic machinery such as an elevator. It may also separate cases according to size or shape by operating an assorting apparatus. Thus, the same equipment may be used for many purposes, depending on what it is used to control,

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Recent developments in the field of television and electronics resulted in a great demand for photoelectric devices, and today the electric eye is looked upon as a dependable and invaluable device for industrial and commercial applications. As men in industry and business realize the value of electronic control, more of these devices will find their way into everyday use. Only the imagination of man stallation, so you can understand better the important part played by the light-sensitive devices which you will study. The six important basic parts of a complete photoelectric control installation-are:

1. The Light Source. The light which is directed upon the electric eye may be the natural light from the sun or artificial light from an incandescent lamp, gas flame, arc light, etc.

2. Light Beam Apparatus. On some photoelectric installations, it is necessary to concentrate the light into a narrow beam in order to make it travel over a definite path before



FIG. 1. Simplified diagram of a complete photoelectric installation such as might be used to protect valuable jewelry on a display table. The infra-red light filter on the light source makes the light beam practically invisible.

stands in the way of accomplishing deeds which are best called *magic*.

► As a sideline for the radio man, the field of photoelectric control offers great opportunities, for, in this branch of electronics, there are many simple, basic applications requiring only standard equipment, a knowledge of the fundamentals of photoelectricity and radio circuits, plus mechanical ingenuity and common sense.

A COMPLETE PHOTOELECTRIC INSTALLATION

Before taking up the different types of light-sensitive cells, we will describe briefly a complete photoelectric inreaching the light-sensitive cell; lenses and curved mirrors are used at the light source to accomplish this. Again, it may be necessary to change the direction of the beam of light with a mirror, or to make the light beam invisible to the human eye by using filters which absorb the visible light rays. Where insufficient light reaches the light-sensitive cell, it may be necessary to use a collecting lens which gathers light and concentrates it upon the relatively small area of the lightsensitive cell.

3. The Photoelectric Cell. The electric eye or light-sensitive cell changes its electrical characteristics in response to changes in illumination.

4. The Photoelectric Amplifier. With certain types of light-sensitive cells, it is necessary to build up the strength of the variations in current or voltage from the light-sensitive cell by means of a vacuum tube amplifier, which may contain one or more ordinary radio amplifier tubes or gaseous tubes.

5. Sensitive Relays. Thephotoelectric cell or amplifier output is used to control some circuit which may be power operated, so a relay is used to close (or open) the power circuits. When the relay is connected directly to the output of the light-sensitive cell, a super-sensitive relay is needed; when connected into the plate circuit of the amplifier tube, an ordinary sensitive relay is satisfactory. Relays which operate on currents of less than 250 microamperes are classed as supersensitive; those which require from .5 to 100 milliamperes (at low power) are classed as sensitive relays.

6. Heavy-Duty Relays. An additional relay, used after the sensitive relay, is necessary in installations where the preceding relay is not capable of handling the current required by the device being controlled. In some very large installations two or even more power relays, one operating the other, are required. Relays which require more than 100 milliamperes are considered the heavyduty type.

► The simplified diagram in Fig. 1 gives you the relations between the various parts in a typical photoelectric installation (the alarm gong sounds if the light beam is intercepted at any point along its path). Fig. 2 shows a typical commercial photoelectric unit having several of the basic parts mounted in one housing.



Courtesy General Electric Co.

FIG. 2. A typical commercial photoelectric assembly. The upper picture is one of the housing, while the lower gives a view of the interior, showing an amplifier tube, the photocell, and a relay. This unit may be used in many ways, requiring only a light source to form a complete installation.

By properly choosing circuits and relays, you can make a change in cell illumination produce any desired control operation, choose the degree of light intensity at which the relays will operate, and speed up or slow down the action of your controls as much as you desire. The only actual limitations to a photoelectric control system are the sensitivity of the light-sensitive cell with its associated apparatus and the ingenuity of the control engineer.

► An example of how this photoelec-

tric equipment operates is the installation in which a photoelectric eye is used to open the doors of a garage when a car enters the driveway. The light source can be mounted on a post on one side of the driveway: this source throws a beam of light across the driveway at such a height that the beam will be intercepted by a car coming in or going out. The beam of light is directed on a light-sensitive cell mounted on the opposite side of the driveway. The apparatus is so connected that nothing happens while the beam of light illuminates the lightsensitive cell. When an automobile approaches, the light beam is interrupted: the light-sensitive cell detects this immediately and causes the value of the current in the plate circuit of the amplifier tube to change. This operates the sensitive relay; its contacts close and send current through the heavy-duty or power relay. When the contacts of the power relay close, current flows through the electric motor which operates the door-opening mechanism. All this happens so quickly that the garage doors are completely open by the time the car reaches them.

WHAT IS LIGHT?

The importance of *light* in any

photoelectric installation should be quite obvious from what has been said up to this time. Let us, therefore, start your study of light-sensitive cells by first learning just what light is and how it is measured.

The greatest source of light is the sun; it sends out waves which are identical with those which we use in radio communication, except that they are a great deal shorter in wave-Light waves which can be length. seen by the human eye vary in length from 40- to 70-millionths of a centimeter. The wavelength of light can, for convenience, be expressed in millimicrons, units of length equal to onethousandth of one-millionth of a meter or one ten-millionth of a centimeter. The human eve, therefore, responds to a light between 400 and 700 milli-Radio waves, which range microns. from .01 to 25,000 meters in length, are more than one million times as long as visible light waves.

The wavelength determines the color of the light; a combination of wavelength (colors) produces the sensation of white light. The human eye does not respond to all colors equally; it is most sensitive to yellow and green. Study Fig. 3 carefully, noting how the human eye responds to the different colors in the visible spectrum.



FIG. 3. The relative sensitivity of the average human eye to light of various colors (various wavelengths) is given by this curve.

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World Radio History

The electric eye, in addition to "seeing" those frequencies of light which can be detected by the human eye, will respond also to <u>ultra-violet</u> light and infra-red light, both of which are invisible to the human eye. It is this characteristic of the electric eye which makes it possible to use invisible light beams to control machinery or to operate burglar alarms.

Light-sensitive cells respond to various types of artificial light as well as to natural light. The ordinary incandescent electric lamp is the most common artificial source of light for photoelectric equipment. Its filament, a very fine wire of high resistance, is heated by a current of electricity until it becomes incandescent and gives off light. Electric lamps designed for automobile headlights are ideal for photoeletric work because the source of light is concentrated into a very small space, approximating a point source of light. The smaller the source of light, the easier it is to focus that light into a beam.

Other artificial sources of light include natural gas lights, coal gas lights, the carbon arc lamp, the mercury vapor lamp, gaseous conduction tubes (better known as neon tubes), and fluorescent lamps.

How Light Is Measured. As you know, the wax or tallow candle was one of the first artificial sources of light. When new light sources began to take the place of the candle, it was natural that their power should be expressed in terms of the old and familiar candle. Eventually a candle made according to certain specifications and burned under certain conditions was selected as the unit of light intensity, and the light given off by this candle was said to have an intensity of one candlepower. If an electric bulb was found to be 40 times as strong as this candle, it was said to have a candlepower rating of 40. The average electric lamp used in the home has an intensity of about one candlepower per watt of power used.

You may have noticed that the strength of the light on a certain object, such as a book, decreases very rapidly as the book is moved away from the source of light. Actually the intensity of the illumination produced by a source of light varies *inversely* as the square of the distance from the source of light. In other words, if the illumination at a point two feet from the source is a certain value, the illumination at a point four feet from the source (twice as far away) will be one-fourth of that at the first point.

The practical unit of illumination is the *foot-candle*. This is the *intensity of the light on a surface* which is directly facing, and one foot away from, a light source which has an intensity of one candlepower. For example, the correct illumination for general reading purposes is about 15 foot-candles; this means that the illumination on the printed page should be equivalent to that which would be produced by 15 standard candles located one foot above the page.

Another unit of illumination, the *lumen*, represents the *amount* of light (or light flux) falling on a surface one square foot in area, every point of which is one foot from a point source of light having a strength of one candlepower. Practically, however, to determine the amount of light (number of lumens) falling on a uniformly-illuminated surface of limited area, you multiply the area of the surface (in square feet) by the intensity of the illumination at the surface (in foot-candles). The answer will be in lumens. To determine the

number of lumens of light emitted by a lamp, multiply the candlepower rating of the lamp by the number 12.6.

The foot-candle is a measure of the intensity of light at a certain point, such as at the face of a light-sensitive cell; the lumen is a measure of the amount of light falling on a given area, such as the sensitive surface of a light-sensitive cell. Light intensity at the cell (in foot-candles) multiplied by effective area of the cell (in square feet) gives the number of lumens of light on the cell.



Courtesy Weston Biectrical Instrument Corp.

This type of light meter is used to measure the illumination in a home or an office. If the amount of light is too low, corrective measures can be taken and the results checked by the light meter.

Brightness is another factor which must be considered by the photoelectric engineer, as he will often direct a light on a certain surface and use an electric eye to pick up the light reflected from that surface. Brightness can be expressed in candles per unit area; this means that brightness is a measure of the ability of an illuminated surface to act as a source of light. For example, a surface has a brightness of 10 candles per square foot when each square foot of this surface reflects as much light in a given direction as would 10 candles.

Light Sources. In most photoelectric applications, it is very desirable to have a definite beam of light directed on the electric eye. Naturally, it is desirable to use as small a light source as possible, in order to economize on power and make a compact unit. Most commercial light sources for photoelectric work use the small but powerful 32-candlepower automobile headlight bulb (the required low voltage is obtained from a step-down transformer which is connected to a 110-volt a.c. line), concentrating the light into a beam with a lens. Since only that light which falls on the lens is useful in producing the beam, reflectors are generally used back of the bulb to reflect light to the lens. Some light is absorbed at each reflection and at each passage through a lens, so it is usually necessary to make adjustments of the light source and the relay apparatus until satisfactory operation is secured.

TYPES OF CELLS

Although photoelectric or light-sensitive cells of various types are being manufactured today by many different firms, all of these can be divided into four basic classes. These four classes of light-sensitive cells are:

1. Photoemissive Cells. In these cells, electrons are emitted from the cathode of the cell by the action of light, and are collected by an anode which is at a positive potential. Photoemissive cells are better known as *photocells*; some technicians prefer to call them phototubes, since they are always built into glass envelopes as are glass type radio tubes. In many technical books and articles you will find all types of light-sensitive cells referred to as photocells, but you can



Concision General Alectric Co.

A wall-mounted photocell assembly. This type is particularly designed to turn on lights in a room when the illumination falls to an unsatisfactory level, and to turn off lights when the level rises above the preset value. The two controls are used to set these upper and lower limits.

generally determine which type of cell is meant from the nature of the discussion. Remember that when we speak of photocells in this book we are referring specifically to <u>photo-</u> emissive cells.

2. Photoconductive Cells. In these cells the resistance (or conductivity) of a material changes under the action of light. Selenium is the resistance material most commonly used in these cells. You will find that photoconductive cells which use selenium are often referred to as *selenium cells*.

3. Photovoltaic Cells. These cells develop their own voltage under the action of light. They are referred to often as photoelectric cells, but photovoltaic cells or self-generating cells are the terms which will be used in this Course. The term photoelectric cell, as used in this Course, will refer to light-sensitive cells in general, regardless of their type.

4. Electron - Multiplier _ Cells. These are really a kind of photoemissive cell, but are so different from the usual photocell that we should treat them as a separate type. An electron-multiplier cell has one cathode and a number of anodes, each of which is maintained at a different positive potential. Through a series of actions, which we will describe later in this lesson, this cell is able to produce many more electrons than are driven out of the cathode by photoelectric action. In this Course, we shall refer to electron-multiplier cells simply as *multiplier* cells.

Each light-sensitive cell has its own characteristics, and naturally transfers these characteristics to the associated photoelectric apparatus. Some cells are more dependable than others; some require external sources of voltage which may fail: some cells have a comparatively short life—all these factors must be taken into consideration where failure of the photoelectric control would in any way endanger human life or damage valuable equipment. We are going to devote the remainder of this lesson to detailed descriptions of each type of cell, telling how they are constructed, how they operate, and basically, how they are used. When you have finished, you will have all the basic information you need to understand the operation of modern photoelectric equipment.

Phótoemissive Cells

The photoemissive cell can be compared to a diode or two-element tube. both contain two electrodes for mounted in a glass envelope. In the radio tube, the electrons which make up the current through the tube are produced by heating the cathode (filament); but, in the photoemissive cell, the light rays activate the sensitized surface of the cathode and cause electrons to be emitted. The other electrode, called the anode, attracts the emitted electrons, as it is positively charged with respect to the cathode. The emission varies with the amount of light: the more light there is shining on the cell, the greater is the emission. Current will flow through the tube; the amount depending on the emission and the applied voltage. The tube acts like a variable resistance, the resistance value following the light variations.

The construction of a typical photocell is shown in Fig. 4. Here the cathode is a semicircular cylinder of metal (usually oxidized silver) supported inside the glass envelope by stiff wire leads made of nickel. The anode is simply a nickel rod or wire placed in the axis of this cathode cylinder. The anode is untreated, but on that surface of the cathode which faces the anode is a thin film made up of caesium oxide, sodium, potassium, or lithium. For practical purposes. just consider the cathode film as simply a layer of light-sensitive chemical compound.

Soda-lime glass is used as the envelope for most of the inexpensive photocells made today, since it is cheap and easy to form into the desired shape. Higher-priced cells use either pyrex glass or fused quartz, as these have lower light losses and allow more ultra-violet rays to pass. Lead glass is never used for pho<u>tocells. It</u> combines chemically with the materials used on the cathode, discoloring the envelope, and lead glass is a poor transmitter of light (it absorbs a high percentage of the rays).

► The modern photocell is made by automatic machinery in much the same way as radio tubes are made. The proper chemical is sprayed on the cathode, the two electrodes are mounted on the glass stem, and then



FIG. 4. The connections and parts of a photoemissive cell.

this stem is sealed to the glass envelope. The gas is evacuated by vacuum pumps, and the tube is given a special treatment in a high-frequency electrical furnace to complete the processing of the coating on the cathode.

Notice that the anode in a photocell is the *smallest* electrode; this is necessary so that the anode will not cast too large a shadow on the cathode and reduce the efficiency of the cell. In regular radio tubes the opposite holds •true; here the plate or anode is the *largest* electrode.

► Two types of photocells are in common use, differing only in that one contains a gas while the other has a "hard" or high vacuum. Gas cells give an increased current output for a given amount of light. The sensitivity of a photoemissive cell is controlled by the cathode materials; the introduction of a gas reduces the effect of the space charge and permits greater sensitivity to be attained.

Since the current passed by the cells is too small to operate even a super-sensitive relay, photoemissive cells are always used with one or more amplifier tubes. A typical circuit which can be used either with gas or vacuum type photocells is given in Fig. 5. The photocell is connected so that the voltages of the B and C batteries will be applied to it through R. (Its current will not operate the relay.) An ordinary radio tube such as the 1H4G (or any triode, provided the proper voltages are applied) serves as amplifier tube; the relay is a sensitive type, with its contacts connected into the power supply leads of the device being controlled.

In this basic circuit the grid of the amplifier tube has a negative bias, and therefore draws no current. The relay resistance (usually somewhere between 1000 and 10,000 ohms) is comparatively low with respect to the circuit resistance, so we can consider the photocell and resistor R to be in series across points A and B of the batteries. When the photocell is dark, (no light on it) there is no emission, so practically no current flows through it and R. Hence, the voltage across R is very low, and the grid of the tube can be considered to have almost the same negative potential as point A.

With a highly negative grid, little or no electron current passes from cathode to plate in the amplifier tube, and the relay armature is not attracted to the relay core.

When light falls on the photocell, its emission rises and more current can flow through R and the photocell, so the voltage across R becomes greater. Electron flow is from A through R,



FIG. 5. A basic circuit which can be used with the photoemissive cell.

through the photocell and through the relay to B; point C is therefore more positive than point A. The grid thus becomes more positive than point A when light falls on the photocell. With a lowered negative bias on the grid, plate current flows through the relay, causing the armature to be attracted to the relay core. In a practical circuit the grid bias would be controlled by a potentiometer connected across the grid bias battery, which would allow the circuit to be adjusted for maximum response to the desired changes in light.

It is only necessary to interchange the positions of R and the photocell in this circuit in order to make the relay operate when the cell is darkened (as when some object interrupts a light beam directed on the cell). In this reversed circuit, light on the cell produces a high current, but the voltage across the low cell resistance is small and is "cancelled" by the high grid bias. Darkening the cell raises its resistance so much that most of the voltage drop across the combination of R and the cell is developed across the cell. This makes the anode of the cell more positive than it is when the cell-is-illuminated; thus, the negative bias of the grid is reduced, and the increased flow of plate current closes the relay.

VACUUM TYPE PHOTOCELLS

In the vacuum photocell the total tube current is made up of electrons that are emitted by the cathode (see Fig. θ). This cell will respond to light variations of almost any frequency, which means that it is ideal for use in the fastest of counting jobs. The capacity between the anode and cathode is the only important limitation on the highest light variation frequency to which this cell will respond.

Just as curves are used to illustrate the characteristics of the vacuum tubes which you have studied, so can curves be used to illustrate photocell characteristics. The stronger the light the greater will be the current flow, but does this relation hold at all times? Is the current affected by the voltage applied to the cell? These are important questions which can be answered easily by making some simple tests.

A standard incandescent electric lamp is used in securing characteristic curves for all types of light-emissive cells, and the voltage applied to it is adjusted to operate the filament at a constant temperature. The illumination on the cell is varied by changing the distance between lamp and cell rather than by varying the light source.

Current-Illumination Curve. This important photocell characteristic curve is obtained by applying a fixed voltage (say 40 volts) to a cell and measuring (with a microammeter) the \cdot current passed by the cell at different intensities of illumination. With a vacuum type photocell, the results obtained will give a curve like that in Fig. 7A, which shows that the current output is directly proportional to the amount of light falling on the cell. That is, the cell current doubles when the light flux is doubled.

Current-Voltage Curve. If a vacuum type photocell, illuminated by a constant light intensity of about .5 lumen, is connected to a variable source of d.c. voltage, and measurements are made of the current passed



FIG. 6. Here are the details of operation of the vacuum type photocell.

by the cell for each value of voltage, the curve obtained will be like that shown in Fig. 7B. Notice that once the voltage reaches a certain value (above the knee of the curve) the cell current increases very little as the voltage is further increased. At this point, called the saturation value of current, practically all of the electrons emitted by the cathode under the action of the fixed light, are drawn to the anode. Clearly, then, there is little to be gained by increasing the voltage above this value. A small size vacuum cell can safely withstand up to 500 volts; often such a high value must be applied to the cell because of circuit conditions.

The curve in Fig. 7B can be made to show the relation between the sensitivity of the cell and the cell voltage simply by dividing the current values by the number of lumens of light on the cell. The vertical scale at the right in Fig. 7B is obtained in this way, by dividing each value of current in the scale at the left by .5, which is the number of lumens of light on the cell. You can easily see that a vacuum type cell should be operated at a voltage corresponding to a point above the knee of the curve if greatest sensitivity is to be obtained

Color Response Curve. Another very important photocell characteristic is its response to light of different colors (wavelengths). The curve in Fig. 7C gives this information for a typical vacuum type photocell. In general, the color response of a photocell differs considerably from that of the human eye (shown by the dashdash curve).

These color response curves are important when the job is one of assorting objects by color; filters must frequently be used to correct the response so that the proper selection will occur, or another cell with a more favorable response must be used. Various lightsensitive materials are used on the cathode to get certain desired color characteristics. Quartz must be used for the envelope where a tube is to be highly sensitive to ultra-violet light, since ordinary glass does not transmit ultra-violet rays.

The characteristics shown in Fig. 7C show a peak in the response in the infra-red light region, so this cell is an excellent one to use with invisible light beams in burglar alarms, etc.

GAS TYPE PHOTOCELLS

The gas type photocell differs from the vacuum type only in that, after



FIG. 7. Typical characteristic curves for vacuum type photoemissive cells. (These curves are for the Westinghouse SR-50 photocell.)

the cell is evacuated, a small quantity of an inert gas (a chemically inactive gas which will not combine with the tube elements) such as argon, helium, or neon is admitted before sealing off the tube. In this gas type cell the electrons emitted by the cathode, traveling at high speed towards the anode, have sufficient force to knock out electrons from gas atoms with which they collide. Thus, they split up the atoms into positive ions and free electrons; this process is known as *ionization*. The electrons resulting from these collisions are attracted to the anode along with the emitted electrons, and serve to build up the photocell current. The positive ions (atoms from which electrons have been knocked out) move to the space cloud near-the-cathode,-neutralizing-the electrons there and allowing more of the electrons from the cathode to find a free path to the anode. If the ionization is made too strong by excessive anode voltage, a glow discharge similar to that in neon sign tubing will be produced. Then the gas ions will bombard the cathode and destroy the tube.

This "ionization by collision" process is shown in Fig. 8. The original tube current can be increased as much as ten times by the ionization of the gas. However, the amount of ionization increases rapidly as the voltage is increased, so the voltage must be kept below a critical value in order to prevent a glow discharge from destroying the tube. With too-high voltages the gas type tube will pass current even when no light is falling on the cathode. The operating voltages vary between 25 and 100 volts for the average gas photoemissive tubes, depending upon their construction. Under no conditions should this voltage be exceeded.

Characteristic curves for a typical gas type photocell are given in Figs.







FIG. 9. Compare these gas cell curves with those of the vacuum type in Fig. 7. (These curves are for the General Electric PJ-23, which is typical.)

9A, 9B, and 9C. The current passed by the gas type cell increases practically in direct proportion to the *illumination*, just as in the vacuum type cell (Fig. 7A).

The curves in Fig. 9B give you a very good idea of the characteristics of a gas type photocell. At voltages below 20 volts this cell behaves almost exactly like a vacuum type cell (Fig. 7B) because, at these low voltages, the emitted electrons do not reach a sufficiently high speed to knock electrons out, of the atoms of gas inside the photocell. At some voltage above about 20 volts (above the knee of the curve) ionization starts, and further increases in voltage give much greater increases in current than are obtained with a vacuum type cell. The dotted lines (Fig. 9B) show how the current would vary at higher voltages if there

the tube to limit current to a safe value in case the illumination is accidentally increased.

Gas Ratio. In the gas photocell, the ratio of the current passed at its maximum safe operating voltage, to the current passed just before ionization (and gas amplification) begins, is known as the *gas ratio* of the tube.





were no gas in the tube.

The voltage and current ratings specified by manufacturers for gas type tubes must be carefully followed if the tube is to be in operation for long periods of time. Voltages slightly higher than rated values shorten tube life considerably. In general, the maximum safe operating voltage of the average gas type photoemissive cell is about 100 volts. It is a good idea to place a resistor in series with For example, the gas ratio in Fig. 9B is about 7; this value is obtained by dividing the current at 90 volts by the current at 25 volts, the illumination being held constant at a value which limits the current at 90 volts to a safe value. The values of illumination used are generally specified for gas ratios, since the ratio depends somewhat on the illumination (it becomes less as illumination on the cell is decreased).

Color Sensitivity. The color

response of a gas type photocell depends upon the kind of glass used for the envelope and upon the nature of the light-sensitive material used on the cathode. The curve in Fig. 9C is therefore not a general response, but is instead the color characteristic associated with the cell used in making this curve and others like it. As you can see, this particular tube has high sensitivity to ultra-violet and infrared, but comparatively low sensitivity to visible light.

GAS AND VACUUM PHOTO-EMISSIVE CELL RATINGS

Just as radio tubes have certain definite ratings which must not be exceeded, so do photocells have maximum voltage and current values which cannot be exceeded in ordinary practice.

The maximum anode current is the maximum value of current which can be safely passed by the tube. For a.c. this is the peak value of current.

The maximum anode voltage is the maximum value of voltage which can

safely be applied to the tube. For a.c. this is the peak value of voltage.

The maximum illumination is more important in connection with gas type cells than with vacuum type cells. At high values of illumination, the voltage applied to a gas type photocell must be considerably lower than the rated value in order to keep the current down to a safe value (in the safe operating range shown in Fig. 9B).

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The sensitivity of a photocell is generally expressed as the current passed in microamperes per lumen of light flux. This is usually measured at a light intensity of either .1 or .5 lumen, so that sensitivity ratings of various tubes can be compared. The sensitivity of vacuum type cells varies from about 5 to 35 microamperes per lumen, while for gas cells, rated sensitivity values may be as high as 300 microamperes per lumen.

Fig. 10 shows a number of typical photoemissive cells. Gas and vacuum type photocells look the same, since the gas used is invisible.

Photoconductive Cells

The photoconductive cell is essentially a resistance whose ohmic value varies with the light falling upon the cell. The stronger the light falling on the cell, the lower becomes the resistance. There is no "emission" of electrons here; the cell is just a resistance which varies with light. When placed in a circuit with a voltage source, the current flow will depend on the voltage and the resistance value at the moment. When dark, the cell resistance is high but it still will pass some current. Hence, there is no "cut-off" of current; it is just varied from a low value to a higher value by the cell resistance variation.

The photoconductive effect was noticed first by an engineer named Willoughby Smith, who, while stationed in the Azores Islands in 1871, observed that the selenium resistors he was using changed their resistance when exposed to sunlight. Reporting the discovery to a group of scientists, he announced: "By the aid of the telephone I heard a ray of light fall on a bar of metal!"

Photoconductive cells are made almost exclusively from selenium, although certain other compounds, one of which is thallium oxysulphide, show appreciable photoconductive effects. Many different forms of selenium cells are on the market today.

► The term "cell" is somewhat misleading when used in connection with photoconductive cells and photoemissive cells, as these do not generate their own voltage and are not therefore to be compared to a battery. The word *cell* is, nevertheless, commonly used among technical men for all classes of light-sensitive devices, and is used as such in this lesson. You will be able to tell what is meant by referring to the circuit diagram, because the symbols for light-sensitive cells are quite different from the symbols for batteries.

SELENIUM CELLS

A selenium cell consists essentially of two electrodes between which is deposited a thin film of selenium. The electrodes, which can be copper, iron, nickel, aluminum, silver, gold, platinum, metal alloys, lead, graphite, or carbon, are mounted on some insulating block such as quartz, glass, clay



FIG. 11. This sketch shows the basic construction of the selenium cell.

compounds, porcelain, slate, mica, or bakelite.

The construction of a simple selenium cell is pictured in Fig. 11. Two pieces of metal foil are cut out as shown, and cemented to the flat base, after which a thin layer of molten selenium is spread evenly over the foil plates and all gaps between the foil. Sometimes the selenium in powder form is sprinkled over the plates, then heated and spread out, or selenium is heated to its boiling point and the vapors allowed to condense on the plates. In any event, the cell must be annealed by heating carefully until the selenium changes from its pitch black form to the gray crystalline form which is highly sensitive to light, then the cell is cooled slowly.

► Commercial selenium cells are inade in a number of different ways. One type of coll has two wires wound over an insulating slab, with a thin layer of selenium deposited between the wires. Each wire serves as a terminal of the finished cell, the resistance between the two wires being determined by the resistance of the selenium layer. A flat piece of slate is sometimes coated with a film of graphite which is polished with a chamois skin, then divided into two interlacing sections with a sharp tool and selenium deposited over the entire surface.

In other types of selenium cells a thin film of platinum, gold, or silver is fused into the surface of a block of glass or quartz. Then, this thin metal film is divided into two sections by a zigzag or comb-like line made with a precision instrument known as a "dividing engine." There may be as many as one hundred lines or scratches per inch on the cell. When the metal film has been divided into two separate electrodes in this manner, a thin layer of selenium is placed over the entire unit, this selenium serving to bridge the gaps between the two electrodes. This method gives a very small separation beween the two plates and a long gap covered with selenium.

► The cell whose characteristics are given in Fig. 12 belongs in this last class; it is highly sensitive to changes



FIG. 12. The Acousto-Lite Laboratories 50A2 selenium cell was used in making these curves.

in illumination but is not recommended for operating voltages higher than 80 volts or light intensities of more than 20 foot-candles. As a precaution against breakdown from excess voltage or excess current, a 1-megohm resistor is usually connected in series with the cell.

By looking first at the Current-Illumination Curve (Fig. 12A), you can see that the current passed increases with illumination, but the effect of the light (the increase in current) is less at the higher values of illumination. Notice that the current does not drop to zero when the cell is dark (the applied voltage being held constant). Selenium cells have a definite "dark" resistance, so current can drop only to the dark level when light is cut off from the cell. The resistance of this particular cell is about 10 megohms in the dark and about 2 megohms at an illumination of 10 foot-candles, hence it has what is called a dark-to-light resistance ratio of 5.

When illumination is held constant on a selenium cell and the voltage is varied, the current varies exactly as it would for an ordinary fixed resistance. You know that the current passed by a resistor varies directly with the voltage applied; that is why the curve in Fig. 12B is a straight line.

You will find that most selenium cells have a maximum sensitivity to red or infra-red light, with a lower peak (in some instances) in the ultraviolet region (Fig. 12C).

Photoconductive cells are generally used with vacuum tube amplifiers, just as are photoemissive cells, but it is possible to make them with a resistance sufficiently low to permit direct operation of a sensitive relay. A battery must be connected in series with the cell and relay, as photoconductive cells do not generate a voltage. The current passed by the cell under constant illumination depends only on the voltage applied to the cell; the maximum current is determined by the maximum voltage which can be applied without causing breakdown of the selenium in the gap between the plates.

The amplifying circuits required for selenium cells are very similar to those used with photoemissive cells. A fundamental selenium cell circuit is given in Fig. 13. Light falling on the

selenium cell changes its resistance, thereby changing the bias voltage on the grid of the tube. In the circuit shown, increases in light make the grid of the tube less negative (more positive), because the cell resistance decreases, allowing a larger current and hence a larger voltage drop across R. This voltage is opposite in polarity to the bias, so it reduces the bias and allows the plate current to increase and operate the relay. With this circuit, therefore, the device being controlled by the relay operates whenever a strong light falls on the cell. (Notice -the response of this circuit to light falling on the cell is similar to that of the circuit in Fig. 5, in which a photoemissive cell is used.) If the positions of the selenium cell and the resistor R are reversed, the relay will be closed when the cell is dark but will open just as soon as sufficient light falls on the cell.

Selenium cells generally are sealed in moisture-proof cases, because they



FIG. 13. Notice the symbol used to represent the selenium cell in this basic circuit.

are quite sensitive to changes in humidity. The stability of the cells is in general rather poor; those cells which have a very high sensitivity to changes in light usually have a short life, and their sensitivity changes with use. Mounting in moisture-proof cases also improves cell stability, but with many cells the current passed varies slightly even with constant illumination. Their power output is limited by the maximum voltage which can be impressed without causing breakdown across the electrodes, and by the amount of heat developed in the cell (excess heat changes the selenium to an insensitive form).

► Selenium cells ordinarily have a time lag and are rather slow to respond to changes in light. It takes several minutes after an increase in



FIG. 14. This H. H. Eby Co. selenium cell is representative of these cells, noted for their small size and for their wide use by experimenters.

light before an ordinary selenium cell will allow maximum current to flow. However, the current reaches 95% of its final value in about .03 second, so the cell is satisfactory for use as a make-break control.

It is possible to change the construction so this cell will respond to light variations (modulations) up to 6000 c.p.s., but the selenium cell is rarely used for audio frequency work.

Advantages of Selenium Cells. Some selenium cells are very sensitive to infra-red light, and so are valuable where control is to be secured with an invisible light beam. Selenium cells can be made to have a low sensitivity to changes in light and a high currenthandling ability, or a high sensitivity with low current-handling ability, whichever is desired. Their output is in general considerably greater than that of photoemissive cells. A typical selenium cell is shown in Fig. 14.

PRECAUTIONS IN USING SELENIUM CELLS

1. Keep the cell cool. The heating effect resulting from: 1, too much current through the cell; 2, from exposure to an intense light source; or 3, from using the cell in locations where temperature is excessive, will cause the selenium in the cell to become soft and possibly melt, which makes it unfit for photoelectric use.

2. Keep the cell dry. When it is not in use, place it in a dark box containing a few pieces of calcium chloride. The chloride will absorb any moisture which may have collected near the cell.

3. Avoid high potentials. Use the

lowest possible voltage which will give the desired results. Choose highresistance relays or place a limiting resistance in series with the cell to protect it. In general, 4 milliamperes are more than ample for relay work.

Exposure to intense lights for long periods of time should be avoided, for this causes "fatigue" which makes the cell temporarily (and in many cases, permanently) insensitive to light. Cells not in use should be kept in the dark, but should be exposed to light regularly, for short periods of time, to aid in retaining their sensitivity. If the resistance of a cell drops greatly, it can be raised, at least temporarily, by applying pulsating or alternating currents.

Photovoltaic Cells

Photovoltaic cells are really small batteries or sources of direct current, since they generate a current which varies with the intensity of the light falling on the cell; more correctly, they transform the radiant energy of light directly into electrical energy. Although the voltage output of these cells is quite low, the current delivered is sufficient to operate an indicating meter or a *super-sensitive* relay without the use of any batteries or auxiliary apparatus.

Types of Photovoltaic Cells. There are two general types of photovoltaic cells: the dry or electronic type, which is today the most common, and the *wet* or electrolytic type. These differ only in the methods of construction and in general characteristics, for each generates its own voltage.

DRY PHOTOVOLTAIC CELLS

This type of cell consists essen-

tially of a metal disc, pernaps 1/16inch thick, on one side of which is a film of light-sensitive material; this basic construction is illustrated at A in Fig. 15. The metal disc forms the positive terminal of the cell, and a thin metal film deposited on part or all of the sensitized surface forms the negative terminal. (This film is so thin it is translucent-that is, it will allow light to pass through it.) The action of light forces electrons to the surface of the sensitized layer, where they are collected by the thin metal film which serves as the negative terminal of the cell. The metal disc must make up for the electrons drawn out of the sensitized layer by the action of light by furnishing electrons to the sensitive layer. This loss of electrons makes the disc become the positive cell terminal. A voltage therefore exists across these terminals when the cell is illuminated, and an electron flow will

be in the direction indicated at B in Fig. 15.

Photovoltaic cell circuits are quite simple. The cell is connected directly to the coil terminals of some type of super-sensitive (meter type) relay, as in Fig. 15C. Since the contacts of this relay ordinarily cannot handle the current required by the device to be controlled, they are generally used to control the current to the coil of a sensitive relay. In some cases, the sensitive relay may be used to control a heavy-duty relay.

▶ While a dry photovoltaic cell can be made in a number of different ways, the following is a typical manufacturing procedure. An iron disc about 2 inches in diameter and 1/16-inch thick is first cleaned thoroughly, then covered on one side with ordinary solder. A thin layer of selenium is deposited over the layer of solder, then the selenium is annealed or heated under



FIG. 15. The construction and operation of the photovoltaic cell differs from that of other cells, particularly in that it generates a voltage without any external battery.



Courses selensum Corp. of America. A typical photovoltaic cell.

pressure. When the cell has cooled, a thin film of gold, silver, or platinum is deposited on the selenium surface, this film being thin enough to allow light to pass through. This film can be applied as a very thin sheet of the metal (called "gold beater's metal") or can be sprayed on the selenium in molten form from a special spray gun. In some cases the translucent metal film is deposited on the selenium by a process called "cathode sputtering." The cell is completed by making contact to the iron disc and to the translucent metal film with thin metal washers of the same diameter as the iron disc. Naturally the cells must be handled very carefully, because the translucent metal film will rub off very readily. A glass window is customarily set into the cell housing to protect the light-sensitive layer.

Although selenium is used as the light-sensitive element, do not confuse these cells with the selenium photoconductive type. The photovoltaiocell generates a voltage, while the photoconductive cell acts only as a variable resistance.

Photronic Cell. The Weston Photronic cell, one of the first commercial selenium-iron type cells to be developed in the United States, is constructed in much the manner described above; today this is one of the best known of all photovoltaic cells in this country.

The component parts of a Photronic cell are shown in Fig. 16; at A is the

housing. The projecting ends of the bolts are of the same diameter as radio tube prongs, and fit into an ordinary four-prong tube socket.

Characteristic curves for the Weston Photronic cell are given in Fig. 17. The current output varies with the external resistance connected to the cell as well as with the illumination; the output is linear (proportional to illumination) for low values of external resistance. This linear relation for



FIG. 16. View of component parts of the Weston Photronic cell, a dry type photovoltaic cell. A—sensitized iron disc; B—glass window; C—metal contact rings; D—bakelite housing; E—terminal bolts; F—bakelite cap; G—assembled cell; H— Photronic cell in weatherproof housing, with visor to keep out unwanted light.

thin disc made of iron, on one side of which is the layer of light-sensitive selenium. The cell is assembled as follows: The glass window is placed in the bakelite housing; the iron disc, with one of the metal contact rings on each side, is placed in the housing next, with the sensitive side against the glass; the contact rings are turned so each terminal lug is over one of the holes in the side of the housing, and the terminal bolts are set into place, heads inside the housing; finally the bakelite cap is screwed into the low resistances holds true even when the cell is in direct sunlight (about 175 lumens of illumination). It is necessary, therefore, to consider the resistance of meters or relays used with Photronic cells. Low-resistance meters are used generally where it is necessary that the current produced be directly proportional to the light intensity. However, for low values of illumination (below .1 lumen or below 10 foot-candles), a sufficiently linear response can be obtained with instruments having 1000-ohm or higher resistances, as the curves are practically straight for illumination of such low intensity (see Fig. 17A).

The voltage output of the Photronic cell for varying intensities of illumination, measured by a method which involved no flow of current through the cell, is shown in Fig. 17B. The curve is not linear, and the values of voltage are quite low. This cell, together with all other photovoltaic cells, is not used generally with vacuum tube amplifiers which require large changes in grid voltage to get useful changes in plate current.

The relative color response curve for the Photronic cell, given in Fig. 17C, shows that the cell and the human eve have a maximum response to about the same color, yellow-green The manner in which a glass window absorbs ultra-violet light can be seen: a much higher response to ultra-violet tight is obtained with a quartz window in the cell. Filters (panes of colored glass) which give the Photronic cell almost exactly the same color response as the human eye can be obtained from the manufacturer, and these filters are necessary whenever the cell is to replace the human eve in making measurements of light. The Weston illumination meter, where the cell is connected directly to a microammeter calibrated to read in foot-candles, is an example of this use.

Photronic cells should be used as current sources rather than voltage sources, because the current output of the cell varies directly with the light falling on the cell. In order to obtain a constant voltage from this type of cell, it is necessary to connect a resistance across the cell and take off the voltage across the resistor. This voltage then will be proportional to the light falling on the cell.



FIG. 17. These characteristic curves are for the Weston Photronic cell, a representative photovoltaic type.

It is possible to connect two or more Photronic cells in parallel to obtain a greater current output for a given intensity of illumination. Connections are exactly the same as for dry cells, minus to minus and plus to plus. The dark resistance of these units is quite high, which means that a number of cells at different points can be connected together in parallel without the possibility of one cell feeding current into the others. The total current delivered by a system made up of a number of dry photovoltaic cells will



Coursesy Dedur-Amsco Corp. A photovoltaic cell is used in this exposure meter. When pointed at the scene to be photographed, the cell operates a meter, which gives information needed for proper exposure of film. This is an important photoelectric application.

be proportional to the total light falling on all of the cells; for instance, if three cells in parallel furnish 100 microamperes when illuminated by light sources of equal intensity, the combination of cells would give exactly 67 microamperes if one cell were completely darkened. Individual cells have this same characteristic; practically the same current response is obtained when light is concentrated on one part of the cell as when the *same amount* of light is spread uniformly over the entire active surface.

The Electrocell. The Electrocell, a dry disc type of self-generating cell imported from Europe, consists of a light-sensitive layer of selenium deposited on the surface of an iron disc. Contact is made to the layer with a transparent conducting film of silver. Transparent lacquer sprayed over the greater part of the cell protects it from rough handling. The sensitivity of this cell is 480 microamperes per lumen; in direct sunlight the 1¾-inch diameter cell will deliver as much as 20 ma. It is claimed that the Electrocell, in the larger sizes, delivers enough current to operate a sensitive relay directly, thus eliminating the need for an expensive super-sensitive relay. The smallest Electrocells, ¾ inch in diameter, have a sufficiently low capacity to be used for modulations up to 8000 cycles.

Other Types. Another type of dry photovoltaic cell which is now on the market, the Westinghouse Photox cell, consists of a disc of copper on which has been formed a thin film of cuprous oxide. Contact is made to the copper and to the oxide layer in much the same way as in the selenium-iron type cell. Gold is the material used as the translucent film on this cell.

► The Lange cell, of foreign manufacture, is quite similar in construction to the above-mentioned photovoltaic cells. According to data furnished by the manufacturer, it has a very good current output.

► The General Electric Company's selenium-on-iron type photovoltaic cell is mounted in a bakelite case and provided with prongs which permit its being mounted in an ordinary four-prong radio socket.

General Characteristics of Dry Photovoltaic Cells. Dry photovoltaic cells usually are connected directly to super-sensitive relays. These are built much like a milliammeter or microammeter, but have contacts on the moving pointer and fixed contacts on the meter scale. Such relays respond to currents of the order of microamperes, and are therefore quite costly.

For all practical purposes, the response of a dry photovoltaic cell to

changes in light can be considered to be practically instantaneous. Actually these cells are fast enough to detect the passage of a rifle bullet through a beam of light. Because of the high parallel capacity of the cell, however, (about .5 mfd.) dry photovoltaic cells are not suitable for such a use as responding to a light beam which has been modulated at audio frequencies. Photoemissive cells are used more generally for this purpose.

A typical assembly of voltaic cells in parallel is shown in Fig. 18. This connection gives a higher output for a



FIG. 18. A number of photovoltaic cells can be connected together as shown here, giving a far greater output.

given light value, as was explained for the Photronic cell.

The output of a photovoltaic cell can be increased considerably by connecting a small potential, not over 6 volts, in series with the cell. Too high voltages may permanently change the characteristics of the cell. In fact, the manufacturers of the Photronic cell recommend that no external voltages be used if maximum cell life is to be obtained. A photovoltaic cell behaves much the same as a photoconductive cell, when an external potential is used in series with the cell.

Dry photovoltaic cells are believed to have an unlimited life; that is, they will retain their characteristics for long periods of time if kept at temperatures below about 120° Fahrenheit. The cells must be tightly sealed in their cases, because they are critically affected by chemical vapors and by dampness.

WET PHOTOVOLTAIC CELLS

The photovoltaic cell in wet form is now over 100 years old, as its principle, known as the Becquerel effect, was discovered by Edmund Becquerel in 1839. While experimenting with the ordinary type of voltaic cell known in that day, he noticed that his cell gave a much greater output in direct sunlight than in the subdued light of his laboratory.

In its simplest form, the wet photovoltaic cell consists essentially of two metals which are immersed in an electrolyte, with one of these metals or electrodes exposed to a source of light: Research workers have developed two types of wet photovoltaic cells, one in which the electrodes themselves are light-sensitive, and another in which the electrolyte is light-sensitive, but neither type is believed to be of great commercial importance at the present time.

Wet photovoltaic cells can be constructed with a number of different electrode materials and electrolytes. One form has two copper electrodes, on one of which is a film of cuprous oxide. Another type uses one copper electrode on which is a film of cuprous oxide, and one lead plate; the electrolyte is a lead nitrate solution.

The wet photovoltaic cell has a number of drawbacks. In some types, destructive gases are formed while the cells are standing idle. Industry in general hesitates to adopt equipment like this, in which there is a possibility of leakage of liquid.

Naturally, polarity must be con-

sidered when connecting photovoltaic cells of all types into a circuit, as the cells are really small batteries. In the wet cell, the electrode which has the oxide layer is always *positive*.

Electron-Multiplier Photocells

A multiplier photocell is a multielectrode photoemissive cell which uses the principle of "secondary emission" to amplify tremendously the electron stream given off from its cathode when the cathode is exposed to light. The RCA 931-A, a typical multiplier cell, has eleven electrodes — the cathode, nine special anodes known as "dynodes," and the regular anode from which the electron stream passes to the outside circuit.

The dynodes are curved metal plates which are coated on one side with a mixture of chemicals. This coating has the property of emitting electrons when a fast-moving electron strikes it. As many as eight or ten electrons may be emitted from the coating when one electron hits it (the number depends to some extent on the velocity of the electron striking the surface). The electrons given off from the coating are called "secondary" electrons, and the emission itself is known as "secondary emission."

HOW THE CELL OPERATES

The elements of the RCA 931-A are arranged as shown in Fig. 19. The dynodes—numbered 1 through 9—are maintained at successively higher positive voltages with respect to the cathode, usually in 100-volt steps. That is, dynode 1 is 100 volts more positive than the cathode, dynode 2is 100 volts more positive than dynode 1, and so on. The dynodes are so shaped and arranged that electrons from one of them tend to go to the dynode with the next higher number —for example, electrons emitted from dynode 2 will go to dynode 3, those from 3 will go to 4, and so on.

As you can see, dynode 9 practically surrounds the anode. This construction is used because the potential between the anode and the cathode will fluctuate with the load voltage drop and such a fluctuating potential would disturb the movement of electrons from one dynode to the next. Dynode 9, however, acts as a shield and prevents this undesirable effect from occurring.

The anode is a grid rather than a solid plate. This construction of the anode allows electrons from dynode \mathcal{S} to pass through it to dynode \mathcal{G} ; then, because of the geometry of the space around the anode, and because the anode is usually maintained at a potential around 250 volts higher than \mathcal{G} , electrons emitted from \mathcal{G} are collected by the anode and passed on to the external circuit.

▶ Let's trace the electron path through this cell to see why high current amplification occurs. Light falling on the cathode causes emission of electrons, just as in the regular photoemissive cell. These electrons are attracted to dynode 1. Each of these electrons, when it strikes dynode 1, may cause the emission of up to eight or ten secondary electrons from the

dynode. These secondary electrons are attracted to dynode 2. where each of them causes the emission of several more secondary electrons. This process is repeated all along the electron path through the tube, with each secondary electron from a dynode causing the emission of many others from the next dynode. As you readily can see, the number of electrons flowing in the tube increases enormously as we go from one dynode to the next. In fact, if the tube worked perfectly, with every electron from one dynode going on to the next and causing the emis-



FIG. 19. The electron-multiplier cell contains a number of elements, so shaped and placed that secondary emission tremendously increases the sensitivity.

sion of ten electrons from it, then just one electron emitted from the cathode would cause one billion electrons to arrive at the anode!

Of course, such perfection of operation does not occur, but the tube can be made to produce current amplifications of as much as 200,000 times which means that 200,000 times as many electrons can be drawn from the anode as are emitted from the cathode. This makes the multiplier photocell by all odds the most sensitive cell in use today.

CELL SHIELDING

The mica shield shown in Fig. 19 is used to prevent possible feedback of positive ions from the anode to the cathode. Although the glass envelope in which the tube elements are housed is as well evacuated as is commercially practicable, a small amount of air is left in it, and some of this air may be ionized by the large electron current in the space between the anode and dynode 9. If such ions were able to flow to the cathode or to one of the first dynodes, they would cause secondary emission from these elements which would, of course, be amplified by the other dynodes of the cell. This would make the electron flow through the cell undesirably erratic, and might change the amplifying characteristics of the cell considerably. The mica shield, which allows no ions to pass through it and emits no secondary electrons when struck by ions, prevents the ions from reaching the cathode or any but the highest-numbered dynodes, and so reduces their effects to a minimum.

► The light shield (see Fig. 19) is a small grill through which light must pass to reach the cathode. It is used principally to prevent undesired light from reaching the cathode. Also, the grill acts as an electrostatic shield, since it is electrically connected to the cathode.

CHARACTERISTICS OF MULTIPLIER CELLS

Since the 931-A multiplier cell is an 11-electrode tube, rather than a diode like other photoemissive cells, we must take ten operating potentials into account when we discuss the performance of the cell. This is not really
as complicated as it looks, since the voltage between the cathode and each successively-numbered dynode usually increases by equal steps. (This is not always true: sometimes the voltage step to one dynode is made unequal to the steps to the other dynodes, in order to control the amplification of the tube.) For ease in discussing the cell characteristics, engineers often



FIG. 20. Sensitivity and gain curves for the electron-multiplier cell.

consider the cell to be an amplifier, with each dynode representing one stage; this allows them to draw characteristic curves in terms of the volts-per-stage used on the cell.

Such a curve is shown in Fig. 20. This shows both the current amplification of the 931-A and its sensitivity in amperes per lumen of light falling on the cathode surface. A comparison of this curve with the current illumination curve for an ordinary vacuum photocell (Fig. 7A) shows you at once how much greater the current output of the multiplier cell is. At an illumination of 1 lumen, for example, the vacuum photocell has a current output of about 12.5 microamperes. The 931-A multiplier cell, operated at 100 volts per stage (a usual operating potential), has a sensitivity of 2

amperes per lumen—that is, an illumination of 1 lumen on its cathode would produce a current output of 2 amperes. The current output of the multiplier is thus 160,000 times as great as that of the photocell for the same amount of illumination!

This, however, does not mean we can draw 2 amperes out of multiplier cell; such a high current would ruin the cell at once. The maximum rated current output of the 931-A cell is 2.5 milliamperes. The high sensitivity



Coursesy ICUA MIN. Co.

I

A typical electron-multiplier cell is shown here.

means we must be sure never to expose the cell to too much light—it might be ruined.

Under normal operating conditions, the output current of the multiplier cell is directly proportional to the illumination. The response of the cell when illuminated by a modulated light beam is perfectly flat up to extremely high frequencies of modulation. The amplifying action of the tube has no adverse effect on its frequency response, because secondary emission from the dynodes occurs instantaneously when they are struck by electrons. At very high frequencies, however, the fact that the electrons take a definite time to move from one element to the next limits the response of the cell. As Fig. 21 shows, the



FIG. 21. Color response curves for the electronmultiplier cell.

color response curve of the 931-A cell has a peak in the ultra-violet region and drops off rather rapidly on either side of this peak.

Power Supply Requirements. Multiplier cells have one disadvantage not shared by other kinds of lightsensitive cells - high voltages are necessary to operate them. The anode may be as much as 1250 volts above the potential of the cathode. Naturally, such high voltages are dangerous, and you should take all possible safety precautions if you work with such a cell. Apparatus using these cells is usually designed so that it is hard to reach the high-voltage electrodes, and so that the primary circuit of the high-voltage power supply must be broken before you can get at the cell. Whenever possible, the positive high-voltage terminal of the voltage supply circuit (connected to the anode of the cell through the load) is grounded; the cathode, connected directly to the negative terminal of the supply, then operates below ground (highly negative with respect This allows the highto ground). voltage section of the circuit to be safely enclosed, while the output section (the part of the circuit you are most apt to touch) will be fairly close to ground potential. Of course, if the circuit happens to be defective, high voltages may appear anywhere in it so you should always shut off the



FIG. 22. Where d.c. is wanted for operation of an electron-multiplier, a circuit like this one can be used.

power and discharge any charged condensers before touching any part of the circuit.

Ordinarily, the operating voltages for the various stages of a multiplier cell are obtained from a tapped voltage divider which is placed across either a full-wave or a half-wave rectified source of current. A simple half-wave power supply circuit for the cell is shown in Fig. 22. A cell also can be run directly from an a.c. source, using a circuit like that in Fig. 23.



FIG. 23. A.C. voltages can be used as the operating voltage for many applications of the electron-multiplier cell.

This circuit is particularly suitable for relay operation.

► A magnetic field, as you know, can

change the direction of an electron stream. For this reason, it is sometimes necessary to shield a multiplier cell magnetically when it is operated near a strong magnetic field, so that the field will not interfere with the desired flow of electrons from dynode to dynode.

▶ In addition to its enormous sensitivity, the 931-A multiplier cell has a low noise level, low dark current, and freedom from distortion. Practical applications of the cell include the operation of relays, sound reproduction from film, facsimile transmission, and scientific research involving low light levels. This cell is a recent development and is now coming into more widespread use.

THE N. R. I. COURSE PREPARES YOU TO BECOME A RADIOTRICIAN & TELETRICIAN OFENISTRATO U.S. PATENT OFFICE (REDISTRATO U.S. PATENT OFFICE)

Lesson Questions

Be sure to number your Answer Sheet 25FR-3.

Place your Student Number on every Answer Sheet.

Send in your set of answers for this lesson immediately after you finish them, as instructed in the Study Schedule. This will give you the greatest possible benefit from our speedy personal grading service.

- 1. Light Source 4. Shothelie a amplificant 1. Report Seam appartus 5. Scrattine Relay & 3. Photoelistic Cell 6 Nearry Bity Relay =
- 1. Name the six basic parts of a complete photoelectric installation.
- 2. What two kinds of light are invisible to the human eye, yet can be "seen" by a photoelectric cell? Whet a Wishert 7 Infoa-Red sufficience Multiplier
- 3. Name the four classes of light-sensitive cells. / Philosofter, Philosofter and a classific cells. / Philosofter, Philosophia cells 3, Photoconductive cells.
- 4. If the photocell and resistor R are interchanged in Fig. 5, will the relay close when: 1, the photocell is *lighted*; or 2, when the photocell is *durkened*?
- 5. What is the maximum safe operating voltage of the average gas type photoemissive cell? 100 U occu
- 6. Which of the light-sensitive cells will operate a super-sensitive relay without the use of a battery or an amplifier? *Photovoltiac Cells*
- 7. What is used with a Photronic cell to make it have exactly the same color response as the human eye?
- 8. How would you connect two or more Photronic cells to secure a greater current output for a given intensity of illumination? In Parallel.
- 9. What basic principle is used in the electron-multiplier cell to obtain a high light-sensitivity?
- 10. How may the amplification of an electron-multiplier cell be varied? From a Lapped Voltage Linder

29

World Radio History

ACTION SPEAKS FOR ITSELF

Be sure you are right, then go ahead—this has been the motto of many of the world's great men.

In most cases you know instinctively what is right, your decision being based upon your past training, your experience, your common sense and your conscience. In these cases, act! Waste no valuable time arguing with others who know less than you; waste no time trying to "pound" your ideas into a cynical world—take the initiative yourself.

It is a thousand times better to *do things* and let your deeds speak for themselves than to spend time explaining why your proposed course of action is right. Friends can hinder your success if you take time to justify your actions to each one of them.

If you need advice—if you are not exactly certain you are right, then go to men who are capable of giving authoritative answers to your questions. You'll find that successful men are glad to answer serious, well-planned questions. Analyze their advice in connection with your own experiences, make your decision, then act!

Give this plan a tryout. You will accomplish a great deal more work, and you will be a lot happier.

J. E. SMITH





Study Schedule No. 26

For each study step, read the assigned pages first at your usual speed, then reread slowly one or more times. Finish with one quick reading to fix the important facts firmly in your mind, then answer the Lesson_Questions specified for that step. Study each other step in this same way.

1. Introduction; Types of Loudspeakers; Bi-Polar Magnetic

A Complete Loudspeaker has Three Sections; The "Weakest Link" in Radio; What You Will Study; four general groups of loudspeakers; Magnetic Loudspeakers; Dynamic Loudspeakers; Condenser. Loudspeakers; Crystal Londspeakers; Headphone Construction; How a Headphone Works; Why Strong Permanent Magnets are Necessary; Headphone Ratings. Answer Lesson Questions 1, 2 and 3.

2. Analysis of Vibrating Systems; Mechanical Resonance

in Headphones

nes - - - - - - - - - - - - Pages 6-11

Electrical equivalent of mass and compliance; Natural Frequency of Vibration; Velocity of a Diaphragm; Maximum Displacement; Velocity of a Vibrating Mass; Facts to Remember About Vibrating Systems; Why headphone diaphragms must have a high natural resonant frequency. Answer Lesson Questions 4 and 5.

3. Balanced Armature Type of Magnetic Driving Unit; Dynamic Driving Units; Condenser Driving Units; Crystal Driving Units - Pages 12-19

Construction of balanced-armature unit; how it works; comparison with other driving units; piston action in a dynamic driving unit; diaphragm construction; how a dynamic driving unit works; voice coil design; factors affecting force upon voice coil; construction of condenser driving units; drawbacks of condenser unit; basic action of a crystal loudspeaker; construction of crystal units; drawbacks of crystal units. Answer Lesson Questions 6 and 7.

4. Loudspeakers Can Also Serve as Microphones; Action of a Horn

as a Loudspeaker Coupling Unit; Design of Dynamic Driving

Units for Horn Loudspeakers - - - - - - - - Pages 19-28

Inter-communicating systems use loudspeakers as microphones; equivalent mechanical circuit of a vibrating diaphragm; Measuring Loudspeaker Efficiency; Loading the Diaphragm; Exponential Horns; Mouth Area; Throat Area; Construction of Horn; Horn Ratings; diaphragm mounting methods; efficiency of horn loudspeakers. Answer Lesson Questions 8, 9 and 10.

5. Mail your Answers for this Lesson to N. R. I. for Grading.

6. Start Studying the Next Lesson.

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World Radio History

HOW SOUND REPRODUCERS OPERATE

Diaphragm-Type Units

Introduction

IN RADIO receivers and in public address systems the loudspeaker serves to convert audio frequency currents into sounds which can be heard by the human ear. The A.F. currents in the last stage of the audio amplifier are made to set in motion an electro-mechanical system, and this in turn causes the air in the vicinity of the loudspeaker to vibrate in accordance with the variations in the audio signal, thus producing sound having all the characteristics of the original sound.

A Complete Loudspeaker has Three Sections. From what you have already studied in this Course and from your own actual experience, you can readily see that a complete sound reproducer or loudspeaker consists of three essential sections: 1. an electromechanical driving unit, often called a motor, which converts A.F. currents into vibratory motion; 2, the diaphragm or cone, which when driven by the motor sets the surrounding air into motion; 3. the air coupling system, consisting essentially of a horn or baffle, which improves the efficiency of the loudspeaker by setting a larger amount of air into vibration and by preventing the reproduced sound components from cancelling each other.

The "Weakest Link" in Radio. Perfection in radio transmitter and receiver design is of little value unless the loudspeaker, the final link in

the radio system, is capable of reproducing the entire range of audio signals with true fidelity and is capable of operating at the required maximum loudness level without distorting the reproduced sound. Contrary to general impressions, the loudspeaker is the weakest link in the entire radio system. The design of high-fidelity loudspeakers is an intricate and involved task even for the engineer who specializes in this particular branch of radio engineering. Even today, after years and years of research on the subject, engineers are still striving for the perfect loudspeaker.

What You Will Study. Although as a Radiotrician and Teletrician you are not particularly interested in the exact details of loudspeaker design technique, a general discussion of some of the problems encountered in building loudspeakers and a study of the various solutions which have been developed for these problems will



The loudspeaker is the mechanical means of converting electrical signals into sound waves.

serve to make you realize the limitations of loudspeakers. Furthermore, a familiarity with the constructional features of various types of loudspeakers and with their performance under various conditions will be of value to you in any branch of radio. Only the more widely used loudspeakers, which you will be likely to encounter in your practical work, will be considered. Basic principles of loudspeaker construction will be stressed, but enough actual examples of commercial loudspeakers will be taken up to illustrate how these principles apply to all loudspeakers.

Types of Loudspeakers

Although some loudspeaker manufacturers have assigned trade names to their various types of loudspeakers, you will find that technical men in general prefer to describe loudspeakers according to their construction or operation.

Loudspeakers can be divided into four general groups according to the type of motor or driving unit which they employ: 1, magnetic loudspeakers, in which the moving element is made of iron; 2, dynamic or moving coil loudspeakers, in which the moving element is a coil of wire; 3, condenser loudspeakers, in which one plate of a large two-plate condenser is the moving element; 4, crystal loudspeakers, in which a crystal is the moving element.

Magnetic Loudspeakers. That type of magnetic loudspeaker which depends for its operation upon the attraction of a bi-polar driving unit for an iron diaphragm has long since been superseded by the other types to be discussed. The bi-polar driving unit is still used almost universally in headphones, however, and will therefore be taken up in this lesson. Magnetic loudspeakers in use today are of the balanced armature type; when a balanced armature unit is made to drive a diaphragm, we have a diaphragm type magnetic loudspeaker. On the other hand, if a balanced armature unit drives a cone, we have a magnetic cone loudspeaker.

Dynamic Loudspeakers. When the fixed magnetic field for a dynamic or moving coil loudspeaker is furnished by an electromagnet, we have what is known as an electrodynamic loudspeaker. When this fixed magnetic field is provided by a permanent magnet, we have a permanent magnet dynamic loudspeaker, often abbreviated as P.M. dynamic loudspeaker. Either of these dynamic or moving coil units can be used to drive a small cone, a large cone or a diaphragm. With a small cone or diaphragm, the air coupling unit is usually a horn; with large cones, of the size commonly found in radio receivers employing dynamic loudspeakers, a baffle ordinarily serves as the air coupling system.

Condenser Loudspeakers. These are also known as electrostatic loudspeakers, for they depend for their operation upon the forces of attraction and repulsion existing between two charged plates. The moving plate is ordinarily about one foot square, so no cone or diaphragm is required. Condenser speakers are used so seldom today that only the fundamental operating principles will be taken up in this lesson.

Crystal Loudspeakers. The driving units of crystal loudspeakers are all essentially the same, but they may be made to drive either a diaphragm

or a cone, and may be used with either a baffle or horn.

Once you become familiar with loudspeaker construction and operation, you will find that practically all of the loudspeaker names or designations which you encounter will be self-explanatory.

Bi-polar Magnetic Driving Units

No discussion of sound-reproducing units would be complete without a consideration of the headphone unit. Aside from the fact that this bi-polar magnetic driving unit was the essential part of all early loudspeakers, headphones are still widely used today by commercial radio operators, by radio amateurs, by experimenters, and by those who are hard-of-hearing.

Headphone Construction. The constructional details of a typical headphone unit are shown in Fig. 1, the important parts being clearly identified. The operating principle of this bi-polar magnetic driving unit can be more clearly understood by referring to the simplified diagram shown in Fig. 2A. The permanent magnet PM, semicircular in shape and made of hard steel, produces a fixed magnetic flux which flows through the two soft iron pole pieces P, and then through two air gaps to the thin iron diaphragm D (about .003 inch thick), which is firmly clamped around its edges to the housing and which is accurately spaced away from the pole pieces. The diaphragm is attracted toward the pole pieces; one explanation is that the magnetic circuit always adjusts itself to have the least reluctance (opposition to the flow of magnetic flux), while another is that the north (N) pole piece makes

the section of the diaphragm directly above it a south (S) pole, and the south pole piece makes the diaphragm section above it a north pole. Since opposite poles always attract, the





diaphragm is bent toward the pole pieces.

The diaphragm is made stiff enough and is originally separated far enough from the pole pieces so that it cannot actually touch the pole pieces under any normal conditions of operation.

How a Headphone Works. Observe that around each of the pole pieces in Fig. 2A is a coil of wire, with the two coils connected together in series in such a way that the magnetic flux due to each will be in the same direction through the permanent magnet. When a current is sent through these coils, it will produce an additional magnetomotive force which either aids or opposes the magnetomotive force of the permanent magnet (depending upon the direction of current flow) and therefore either in-



FIG. 2A. Simplified diagram of a bi-polar magnetic headbhone unit, showing coil connections and flux paths.

creases or decreases the amount of magnetic flux flowing through the magnetic circuit of the headphone. Any change in this total magnetic flux naturally changes the attraction of the pole pieces for the diaphragm; when the current flowing through the coil is varying at an audio frequency, the diaphragm will move toward and away from the pole pieces at the same frequency. With proper design of the entire headphone unit, the movement or displacement of this diaphragm will correspond exactly to the variations in the audio frequency currents. and this diaphragm will produce sound waves which correspond in wave form to that of the A.F. signals.

Why Strong Permanent Magnets are Necessary. One of the most important factors affecting the performance of a magnetic loudspeaker is the permanent magnetic field. Let us see what would happen if this field were not used. Assume that we are sending a pure sine wave alternating current through the coil of the bi-polar driving unit in Fig. 2A, with permanent magnet PMtemporarily replaced by a soft iron piece of similar shape. When this alternating current is zero there will be no magnetic field and no pull on the diaphragm. As the current swings positive, the two poles will become magnetized with opposite polarity, and the diaphragm will be pulled toward the pole pieces. This attraction will continue until the current again reduces to zero. As the current reverses for the negative half of the cycle, the polarity of each pole will reverse, but the poles will again attract the diaphragm. As a result, the diaphragm will vibrate at twice the frequency of the current passing through the coil. Clearly this is an undesirable condition for sound reproduction.

The use of a strong permanent magnetic field eliminates this undesirable frequency-doubling effect in a bipolar headphone unit, for the permanent field provides a continuous attraction for the diaphragm which is merely varied by the alternating current. The change in the attraction of the pole pieces for the diaphragm depends upon the original magnetization and upon the change in flux caused by the signal current flowing through the coils (the change in attraction actually depends upon the product of these two factors). Increasing the strength of the permanent magnet and increasing the strength of the signal currents are thus possible ways of increasing the sensitivity of a headphone unit.

Modern headphone units have permanent magnets made of chrome or tungsten steel, which produce a strong permanent flux and thereby give a highly sensitive unit with a minimum of frequency doubling. Permanent magnets made of Alnico produce even greater magnetomotive forces, and this recently developed alloy may therefore be found in new highquality headphones. Placing the diaphragm as close as possible to the pole pieces increases the amount of permanent flux and thus increases the sensitivity of the unit, but at the same time this procedure reduces the maximum available volume; the practical headphone unit is therefore based upon a compromise between sensitivity and volume.

If the coil current in a bi-polar magnetic driving unit is to produce a large A.C. magnetic flux, many turns of wire are needed on the coils, and the A.C. magnetomotive force produced by these turns must develop a large A.C. flux. You know that in a magnetic circuit the amount of flux produced for a given magnetomotive force depends upon the reluctance of the flux path, just as the current for a given voltage depends upon the resistance of the path for current in an electrical circuit. The first essential of a low-reluctance flux path is a small air gap between each pole piece and the diaphragm, but on the other hand, the air gap must be long enough

to permit the diaphragm movements which are necessary for maximum required volume. With the arrangement shown in Fig. 2A, however, the A.C. flux produced by the coils must flow through the permanent magnet; unfortunately this has a high reluctance because it is made from a hard, tempered steel or steel alloy rather than from soft iron.

A simple solution to this difficulty is shown in Fig. 2B; it involves placing a short-cut path for A.C. magnetic flux across the bottoms of the



FIG. 2B. Simplified diagram of a special type of bi-polar magnetic headphone unit in which a leakage reactance path R for magnetic flux is purposely introduced to improve performance.

pole pieces. The air gap R which is placed in this path has a low enough reluctance to make the *alternating* magnetic flux prefer it to the path through the permanent magnet, but at the same time this air gap offers considerably more reluctance to the *per*manent magnetic flux than does the desired path through the pole pieces and the diaphragm.

Headphone Ratings. Headphones are commonly rated according to their D.C. resistance, even though the impedance value gives a far better indication of operating characteristics. For a given type of headphone construction, having a given size of bobbin for each coil, we can use a large number of turns of very small wire, thereby securing a high D.C. resistance and a high impedance, or we can use a small number of turns of larger wire to secure a low D.C. resistance and low impedance. Since headphones ordinarily serve as plate loads in amplifier stages, it is the impedance value (generally measured at the avermanufacturers supply this information.

Analysis of Vibrating Systems

A complete study of the mechanically vibrating parts used in loudspeakers involves a number of new fundamental ideas which you should understand before proceeding færther with your study of loudspeakers. A mechanical vibrating system has certain properties which compare with those of an electrical vibrating or



Radio operators for commercial or amateur communication systems ordinarily use a pair of headphones like those shown here, in which two units are mounted on the headband and connected together in series. Single headphones with a headband are widely used by experimenters, by hospitals having radio facilities for each patient, and by the hard-of-hearing. A single headphone unit with a handle is used chiefly by those who have only slightly impaired hearing and who therefore need use the sound-amplifying system only at intervals.

age voice frequency of 1,000 cycles) which is of importance; the designer will choose a headphone impedance value which best matches the required plate load of the amplifier tube. In general, the impedance of a headphone unit at 1,000 cycles will be from five to ten times its ohmic value for direct current.

When headphones are being used primarily to convert low intensity . signals into audible sounds, they should be rated on the basis of how much A.F. power in milliwatts is required to produce a just audible 'sound; as yet, however, very few

oscillating system; for instance, you will learn that an <u>inductance (coil)</u> in an electrical oscillating circuit can be compared to mass in a mechanical vibrating system, capacitance in an <u>oscillating circuit</u> can be compared to *compliance* in a mechanical system, and resistance in an oscillating circuit can be compared to *friction* or to *mechanical resistance*. A few examples will help you to understand what these new terms mean.

Consider any large, heavy body on wheels; an automobile will do as an example. When pushing the car on a smooth, level road, you know that

considerable force is required to set it into motion, but once the auto is moving at a definite speed, it can be kept at that speed with little additional effort. In order to set such a heavy object into motion, we must overcome the inertia of its mass; the heavier the object, the greater is its mass and inertia, and the more we must push to set it into motion. We have the same situation in an electrical circuit: an inductance has an electrical inertia which serves to limit the initial flow of current. Mass in a mechanical vibrating system therefore corresponds to the inductance of a coil in an electrical circuit

Now let us consider an ordinary steel coil spring. Experience has taught us that force is required to stretch the spring, and now we learn that this ability of a spring to stretch under the application of force is referred to by engineers as its compliance. The greater the compliance of a spring, the more it will stretch when a given force is applied. This compares very closely to a condenser, where greater capacity means that we can store more electricity in the condenser.

As you well know, motion of any object involves friction. It is far more difficult to move an automobile from which all four wheels are removed than a car having wheels, for without wheels there is a great deal more friction between the chassis and the pavement. You know also that friction can be reduced by making the sliding surfaces perfectly smooth and by lubricating the sliding surfaces with grease. Friction of the type which exists when one object rubs against another is quite familiar to you, but in loudspeakers it is the friction produced by an object moving in air which is of importance. An example will illustrate the nature of this friction.

Suppose we take a mass, as shown in Fig. SA, and a spring, as shown in Fig. SB, and connect them in series as shown in Fig. SC. When both the spring and the mass are stationary, pull down on the mass and then release it. The spring stretches when you pull down, but since it is under tension it immediately starts to pull the mass up again after you have re-



FIG. 3. These diagrams will help you to remember that mass can be considered as a mechanical inductance, that springiness or compliance of any kind (a spring) can be considered as mechanical capacitance, and that a complete mechanical vibrating system acts in much the same way as a seriesresonant electrical circuit.

leased it. Having reached its original position, the mass continues moving upward because it has acquired momentum. The mass stops moving upward when it has lost this momentum: the mass then drops downward, moving past its normal stationary position again because of momentum. and continues to bob up and down in what we call mechanical vibration. This vibration continues until all of the original energy imparted to the spring and mass by pulling it down is wasted in friction. This friction occurs in two different ways here, between the mass and the air particles, and internal friction produced

by bending of the spring. If this entire system were in water or in a thick oil, there would be even greater friction between the water or oil particles and the mass than in the case of air, and the vibration would stop sooner.

We need not necessarily have fivepound weights and large, stiff springs in order to secure a mechanical vibration; the weight or mass need be only a fraction of an ounce, and the spring can have far more compliance than that shown in the illustration. No matter how small the mechanical structure happens to be, vibration can be secured if there is mass and compliance in the system.

Now you can readily see that the diaphragm in a bi-polar magnetic driving unit has mass and compliance. It will therefore vibrate. Radio engineers have found that its exact behavior during vibration is far easier to understand if the mechanical system is replaced by an equivalent mechanical circuit which has all the appearances of an electrical circuit. Figure 3D shows such a circuit for the diaphragm of the headphone unit; it will make the mechanical action of a headphone considerably easier for you to understand. Remember, however, that this is not the actual electrical circuit in the headphone; it is simply an imaginary series-resonant circuit which acts in the same way as the mechanical system.

Natural Frequency of Vibration. Notice that we have mechanical force $F_{\rm m}$ acting upon mechanical inductance $L_{\rm m}$, upon mechanical capacitance $C_{\rm m}$ and upon mechanical resistance $R_{\rm m}$. When this force is applied for a short instant of time and then removed, the system goes into vibration and me-

chanical current i_m , which you can think of as the velocity at which the diaphragm moves back and forth, is the result. The natural frequency of vibration of the diaphragm (or of any other physical object) will be determined by the mechanical inductance (mass) and the mechanical capacitance (compliance) of the diaphragm or other object. When a continuously vibrating force is applied, the vibration of the assembled parts will follow this force, and the amount of mechanical current will depend upon the mechanical reactance of L_m and C_m





and the mechanical resistance of R_m , just as in any series resonant circuit. By adjusting the frequency of the vibrating force or by changing L_m and C_m mechanically, you can make the mechanical circuit act either as a mechanical resistance, a mechanical inductance, or a mechanical capacitance.

Velocity of a Diaphragm. Let us get a clearer picture of this mechanical current or velocity, for it is extremely important in connection with our study of loudspeakers. Suppose we have a mass and compliance arranged as shown in Fig. 4A; we set this mechanical system into vibration. Under this condition, any particle on this mass, such as point X, will be . When particle X is moving across moving continually above and below the normal line at point 1, it has a its normal position or normal line. If we plotted from instant to instant the position of particle X with respect to this normal line, we would secure the

greater mile-per-hour speed than at any other point along its path; in other words, the particle has maximum velocity when it crosses the



These diagrams illustrate a number of interesting features of horn loudspeakers. A--Low-power magnetic driving unit of the balanced armature type, with diaphragm feeding into a standard $\frac{1}{2}$, inch throat for use with horn loudspeakers; B--balanced armature magnetic driving unit in its housing, with connecting leads clearly visible. Units like this can handle up to 2 watts of input power; C--electrodynamic driving unit for horn loudspeakers. Note the two field coil leads coming out of the side of the housing or pot; D--special three-way connecting throat which permits the attachment of three separate driving units to a single trumpet or horn; E--another electrodynamic driving unit for horn loudspeakers. Size and weight are general indications of power-handling ability in units of this type. A 20-pound unit will handle approximately 25 watts of input power continuously and 10 watts on peaks, while a 6-pound unit will handle about 5 watts continuously and failure may rattle because of excessive disphragm displacement. The knutled protective cap on the throat is removed when the unit is installed; F--dynamic driving unit is a large disphragm which is coupled to a trumpet horn; G--horn loudspeaker designed specifically for high frequencies, from 3,000 cycles to 15,000 cycles. The trumpet horn used on this unit is spun from a single piece of aluminum. These diagrams illustrate a number of interesting features of horn loudspeakers. aluminum.

solid line curve shown in Fig. 4B. Yes, this plotting of displacement against time for a vibrating mechanical system results in a pure sine wave; this is why mechanical vibration of a simple mass and compliance As particle X drops down toward the has a simple sine wave motion.

normal line. A quarter of a cycle later, when particle X is at point 2, its highest position above the normal line, it actually comes to a standstill for an instant, and its velocity is zero. normal line again, its velocity in-

creases again to a maximum as it goes through point 3, and then gradually decreases to zero again at point 4. If we plot this velocity of particle X from instant to instant, we secure the dotted line curve in Fig. 4.

Maximum Displacement. When we deal with the vibrating systems used in loudspeakers, we are particularly interested in the peak or maximum displacement of the mass or of particles on it. In the bi-polar magnetic driving unit, for example, a strong coil current will give a greater displacement than a weak current, and an alternating coil current will cause the diaphragm to move in and out in an alternating manner similar to that shown in Fig. 4B. If the alternating current is of high frequency, then the diaphragm will likewise vibrate at this same high frequency.

Suppose that the diaphragm is moving the same distance (has the same peak displacement) first for a low frequency and then for a high frequency; clearly it will take less time to go through one complete cycle at the higher frequency. The speed or velocity of the diaphragm is therefore considerably greater at the higher frequencies of vibration.

Velocity of a Vibrating Mass. At a given frequency of vibration, any increase in displacement likewise means an increase in the velocity of the mass or particle. Your own experience will prove this to be true, for you know that the farther you pull the weight in Fig. 3C down from its "at rest" position, the faster will it move upward past its "at rest" position when released. The velocity of a vibrating mass therefore depends upon the frequency of vibration and upon the maximum displacement of the mass from its normal position.

When the diaphragm of a headphone or loudspeaker is vibrating, it causes particles of air in its vicinity to vibrate in a similar manner. These air particles will vibrate at the same frequency as the diaphragm, and will also have the same velocity as the diaphragm at any instant. The vibration is transmitted from one air particle to another, giving rise to sound waves. It is the frequency at which



Courtesy Arlavox Mfg. Co.

This magnetic loudspeaker, available with cone diameters ranging from 3 inches to 8 inches, will be found in many small Universal A.C.-D.C. receivers. It uses a balanced armature type of driving unit. Note the U-shaped permanent magnet.

these air particles vibrate which determines the pitch or tone of the sound, and it is the *velocity* of the *air particles* which determines the power * (loudness or volume) of the sound.

^{*}You know that in an electrical circuit like that in, Fig. SD, the power absorbed by the resistor is equal to the square of the current multiplied by the ohmic value of the resistor ($P = I^*R$). Likewise, in a mechanical vibrating system, the square of the air velocity multiplied by the mechanical resistance gives the mechanical power used by the system; this represents acoustical power.

Facts to Remember about Vibrating Systems. It will be some time before you will be able to appreciate fully the facts just brought out in connection with the vibrating systems used in loudspeakers. For your present study of loudspeakers, it will be sufficient for you to keep in mind the following important facts:

- 1. Any physical object has mass, which can be considered as mechanical inductance.
- A physical object can have compliance (you can call this springiness if you prefer), and this characteristic can be considered as mechanical capacitance.
- The motion of any physical object is accompanied by friction, which represents power used or wasted.
- 4. Any physical object which has mass and compliance can be shocked into vibration by a sudden application of force; it will then vibrate at its natural frequency of vibration, which is determined by its mechanical inductance (mass) and its mechanical capacitance (compliance). Increasing either of these two values lowers the natural frequency of vibration.
- 5. The application of a steady alternating force to a vibrating system will cause a corresponding alternating or vibrating motion of the system at the same frequency. We can consider this mechanical vibrating system as equivalent to an electrical series resonant circuit. When the natural frequency of the system corresponds to the frequency of the applied alternating force, resonance occurs and the system acts as a mechanical resistance. At frequencies above resonance, the vibrating system acts as a mechanical inductance (a mass). At frequencies below the natural resonant frequency, the system acts as a mechanical capacitance.
- 6. Since mass can be considered as an inductance, and compliance as a capacitance, these parts of a mechanical

vibrating system have the additional property of mechanical reactance, which takes frequency into account.

7. The application of an alternating force to a vibrating system causes the system to vibrate with a velocity dependent upon its mechanical reactance and mechanical resistance.

Mechanical Resonance in Headphones

Referring again to Fig. 2A, we recognize that the diaphragm is the mechanical vibrating system in this bi-polar magnetic driving unit. Even though quite small, the diaphragm has mass (weight) and compliance (springiness). When the diaphragm is set into vibration by an alternating current flowing through the coils, it causes air particles in the vicinity to vibrate in a corresponding manner. The resulting sound produced naturally represents radiated energy, and therefore the diaphragm must overcome mechanical resistance in radiating this energy. The diaphragm has a small mass and therefore its mechanical inductance is low; being quite stiff. it has a low compliance and its mechanical capacitance is likewise small. The natural frequency of vibration of the diaphragm is therefore quite high; in fact, diaphragms in high-fidelity headphone units are designed to have a natural frequency of vibration which is considerably above the highest audio frequency being reproduced. This prevents resonant peaks in the response curve of the headphone. Clearly the question of mechanical resonance is of great importance when high fidelity is required from a bipolar magnetic driving unit such as is used in a headphone.

Balanced Armature Type of Magnetic Driving Unit

Although the bi-polar magnetic driving unit is remarkably simple in design and can be made to have high sensitivity, it is not entirely suitable for loudspeaker applications where high sound output (high volume) is required. Application of excessive input signal voltage to this type of unit results in distortion, due either to the diaphragm striking the pole tips or to a frequency-doubling effect which occurs when the air gap is increased to permit greater volume. Furthermore, normal movements of the diaphragm cause the air gap to vary, with the result that the permanent magnetic flux also varies and a certain amount of When the diadistortion occurs. phragm is made stiff enough to prevent it from hitting the pole pieces, it then becomes quite thick, and its increased mass lowers the natural frequency of vibration. A stiff, lightweight disc would be desirable, but this cannot be easily obtained in the bi-polar type of construction.

The balanced armature type of magnetic driving unit, illustrated in Fig. 5A, overcomes most of these disadvantages. A light-weight strip of soft iron, called the armature, is pivoted in the center of a coil through which flows the A.F. signal current. This coil may have only a few turns when large A.F. currents are available, but may be built with several thousand turns for low A.F. currents. The coil does not interfere in any way with the movement of the armature. The ends of the armature move between two U-shaped pole pieces which are clamped to a U-shaped permanent magnet which is generally larger

and more powerful than the permanent magnet used in bi-polar units. The permanent magnet in this case makes one set of pole pieces permanently of north polarity and the other of south polarity. One end of the armature is rigidly linked to a diaphragm which may be made of mica, duralumin, or any other light material; if the diaphragm is not sufficiently stiff a spring steel metal strip may be attached to the other end of the armature as indicated, to increase the stiffness (reduce the compliance) of the system.

When no current is flowing through the coil, the armature is not magnetized. All four pole pieces then attract the armature equally, and we have a balanced condition in which the armature stays midway between the pole pieces. It is from this characteristic that the balanced armature magnetic driving unit gets its name. When a current i flows through the coil in the direction shown (arrow indicates electron flow), this current will magnetize the armature and give it the polarity indicated in Fig. 5A. The N pole of the armature will be attracted to the nearest S permanent pole and will be repelled by the permanent N pole. A similar action occurs at the other end of the armature. which is magnetized with south polarity. The result is that the armature twists on its pivot in a counterclockwise direction, as indicated by the curved arrow line, pushing the diaphragm upward. When the current through the coil reverses its direction, the armature is twisted in the other direction and the diaphragm is pulled downward.

Figure 5B shows the two paths taken by the magnetic flux produced

by the armature when currents flow through the coil. These paths are entirely through soft iron, which has a low reluctance and therefore permits a large amount of flux to circulate. Notice that movement of the armature in either direction will cause one air gap in one of the paths to decrease and the other air gap in that path to increase, with the result that the total air gap for any one path remains essentially the same regardless of the position of the armature. This is a desirable condition for distortionless operation.

Now let us see what effect the position of the armature has upon the path taken by the permanent magnetic flux. Referring to Fig. δB , let us assume that the armature is rotated counter-clockwise, so that air gaps g_1 and g_2 are considerably smaller than the other two. Naturally the flux coming out of the north pole of the permanent magnet will choose to take the path through pole N_1 because g_2 offers less reluctance than gs. This flux will pass through the entire armature and then through gap g_1 to pole S_2 . The result is that poles N_1 and S_2 are strengthened. while poles N_2 and S_1 are weakened. But remember that the motion of the armature depends upon repulsion as well as attraction; an increase in the attraction on any one end of the armature is offset by a decrease in the repulsion on the same end, with the result that the effect of the permanent magnetic flux upon the armature is essentially independent of the position of the armature.

A balanced magnetic driving unit can swing through considerably greater distances than a bi-polar magnetic unit without serious distortion. The use of large and powerful permanent magnets made of new magnetic alloys makes possible a large air gap, with resulting increase in output volume. The moving system can be made light enough to prevent mechanical resonance in the range of audio frequencies being handled.

With the balanced armature type of construction there is generally ample



FIG. 5. These diagrams illustrate the construction and principle of operation of a balanced armature type magnetic driving unit.

room for the coil, permitting the use of larger-sized wire with fewer turns in order to reduce the coil inductance. This allows higher-frequency audio currents to flow, increasing the frequency range of the loudspeaker. In general, then, a balanced armature type magnetic driving unit gives higher output and a wider frequency range than a bi-polar magnetic driving unit. With proper design the magnetic type of driving unit can be made to have reasonably uniform frequency response over the middle range of frequencies, from 150 cycles to 3,000 cycles.

Dynamic Driving Units

The amount of audio sound power radiated by a given loudspeaker at a definite frequency depends upon two things: the area of the diaphragm, and the displacement or maximum movement of the diaphragm. These two factors together determine the



FIG. 6. Simplified diagram illustrating the operating principles of an electrodynamic driving unit.

volume of air moved and thereby determine the amount of sound produced at a particular frequency. (The inand-out motion of a loudspeaker diaphragm is often referred to as *piston action*, for the diaphragm acts much like the piston in a water or air pump; the larger the piston area and the farther it moves, the more water or air will be moved.) The greater the volume of air which is moved by a given diaphragm, the greater is the load on the diaphragm and the greater is its mechanical resistance.

With the maximum displacement of the diaphragm limited by the air gap length in magnetic types of loudspeaker driving units, engineers turned to an entirely new type of electro-mechanical. driving system, known as the <u>dynamic loudspeaker</u>. A study of the simplified diagram in Fig. 6 will show the operating principles of this dynamic unit, which consists simply of a small coil, carrying signal current, which is attached to a large diaphragm and which is located in a strong magnetic field.

Bear in mind that the diagram in Fig. 6 represents a cross section of a cylindrical unit. The permanent magnetic field is provided by sending direct current through a field coil wound around the center cylindrical soft iron core. An iron shell or pot surrounds this core and coil to complete the magnetic path up to the air gap in which is located the voice coil; this coil consists of a few turns of fine wire wound on a bakelite-impregnated paper or mica coil form. It is essential that the air gap be kept as small as possible without having the coil or its form rubbing against the cylindrical pole piece or the central core. To reduce the thickness of this voice coil, a spiral groove is often cut in the form for the wires. Mica is often used for the coil form because it will not warp when subjected to excessive heat due to resistance losses in the voice coil. The leads to the voice coil must, of course, be exceptionally flexible and arranged in such a way that they do not interfere with the movement of either the coil or the diaphragm.

-

Diaphragm D, to which the voice coil is attached, is usually made of stiff duralumin, in order that the coil will return to its normal no-current position when the voice coil current drops to zero. The diaphragm is corrugated around its edges, such as at R, in order that the inner portion of the diaphragm can move with the voice coil even though the outer edges are rigidly clamped in position.

Now let us see how this dynamic driving unit operates. First of all, I want to point out that the operation is the same regardless of whether the permanent magnetic flux through the air gap is produced by a coil carrying direct current or by a powerful permanent magnet having essentially the same shape as the soft iron core shown is present in the vicinity of the voice coil at all times.

As you know, any wire which is carrying current will have around it a circular magnetic field like that shown in Fig. 7B. If the electron flow through this wire is into the paper, the magnetic lines of force will have the direction shown.

When this single current-carrying wire is in the air gap, we have the conditions shown in Fig. 7C. The two magnetic fields react with each other. Notice that the magnetic lines



FIG. 7. What makes the voice coil move in a dynamic loudspeaker? These diagrams will tell you. The fixed magnetic field is shown at A, while the magnetic field around a voice coil wire is shown at B. The interaction of these two fields, as at C, results in motion of the voice coil. The + symbol in the wire indicates that the electron current is flowing into the paper (away from you) through the wire.

in Fig. 6. It is the interaction between the permanent magnetic flux and the flux produced by the voice coil current which produces motion; the principle is the same as that applying to electric motors.

To understand just why the voice coil should move when current passes through it, consider only one part of the air gap, as shown in Fig. 7A. The magnetic lines of force here come out of the ring-shaped N pole piece and go through the air gap to the cylindrical center core which is made an Spole by the permanent magnet or by the field coil. This is the field which of force underneath the wire are in the same direction and reinforce each other, while those above the wire are in opposite directions and tend to cancel out. There is a crowding of flux lines below the wire and a less-thannormal number above the wire; the motion of the wire will be such as to redistribute the flux more uniformly, and the wire will therefore move upward as indicated by the long arrow. If the current through the wire reverses, the magnetic field of the wire will be reversed in direction and the wire will be forced downward. With a voice coil in the air gap, this same

action takes place at all points in the air gap and on all'turns of the voice coil, with the result that the entire voice coil is either moved upward or downward depending_upon the direction of current flow.

The strength or magnitude of the force acting upon the voice coil depends upon three things: 1, the strength of the fixed magnetic field existing in the air gap; 2, the length of the wire used for the voice coil; 3, the voice coil current. Increasing any one or all of these three factors increases the force acting upon the voice coil.

If the force acting on the voice coil is to be proportional to the voice coil current at all times, the magnetic flux must be of uniform strength throughout the air gap. You will generally find a groove cut into the central core at U to prevent the flux from thinning out at the lower edge of the air gap; furthermore, the length of the voice coil will generally be less than the height of the air gap.

The weight of the diaphragm and voice coil assembly should be low, and the diaphragm should have sufficient stiffness so that mechanical resonance will occur above the audible frequency range.

Since the voice coil in a dynamic unit has only a few turns of wire (from 1 to 50 turns), its reactance is quite low even for the highest audio frequencies; the high-frequency response of a dynamic loudspeaker is therefore not limited by the voice coil inductance, but rather by the mass of the moving system. Diaphragm movements of one-half inch are not uncommon in dynamic loudspeaker units which are capable of handling large amounts of audio power.

Condenser Driving Units

Although condenser loudspeakers are relatively little used today, they have certain special advantages which make it advisable for you to be familiar with their construction and operating principles. They depend for their operation upon the attraction of a positive charge for a negative charge, and might therefore be called electrostatic units.

In a condenser driving unit a heavy, stationary plate is charged with one polarity and a light, movable plate, mounted a short distance away, is charged with opposite polarity. The separation between plates is uniform and is made as short as possible without causing the charges to equalize by jumping across the air gap. As you know, these two plates constitute a condenser; increasing the areas of the plates and reducing the distance between them increases the capacity of the condenser and thereby increases the charge which can be stored on the plates. For a given electrical charge on the plates, reducing the air gap between them increases the force of attraction between the plates.

Application of a high-voltage A.F. signal to the plates is not enough to give the desired operation, for the plates will always be of opposite polarity and will therefore always attract each other (except when the voltage is zero, which occurs twice during each cycle). As a result, we have a frequency-doubling effect just as would occur in a bi-polar magnetic unit if there were no permanent mag-The remedy is quite simple and net. comparable to that used in the headphone unit; a D.C. or polarizing voltage is applied to the plates, so that

there is a constant attraction between the plates. The A.F. voltage acts in series with this fixed or polarizing voltage, increasing or reducing the attraction. In this way the displacement of the light, movable plate corresponds to the wave form of the A.F. signal.

In practical condenser loudspeakers the movable plate usually consists of metal foil which is cemented over a sheet of thin, stretched rubber which is supported at its edges. A D.C. polarizing voltage of about 500 volts is applied to the plates. The essenlimited life, becoming hard or deteriorating with time. The low-frequency response of a condenser unit is rather poor, and some provision for correcting this must ordinarily be made in the audio amplifier. The high-frequency response is quite good, however, for the reactance of the loudspeaker unit decreases with frequency and the mass of the moving system is not great enough to prevent highfrequency movement. High-frequency response is further improved by punching holes in the stationary plate permit free movement of air to



FIG. 8. Cross-section view of a typical condenser loudspeaker, showing essential constructional features. The plates were usually made about 8" x 12" in size.

tial features of this construction are shown in Fig. 8.

Although condenser loudspeakers are simple in construction and comparatively inexpensive, their power output is limited because of the extremely limited displacement of the moving plate (the maximum obtainable displacement is always much less than the thickness of the stretched rubber sheet). When increased output is desired, it is necessary to use several condenser loudspeaker units arranged side by side and connected in parallel; each unit ordinarily is about one foot square. Another drawback is the fact that the rubber used as an insulating material has a through this plate. Sounds travel out from a condenser loudspeaker at right angles to its moving plate, and the units therefore have desirable directional characteristics.

Crystal Driving Units

The basic action of a crystal loudspeaker is as follows: The application of a voltage of given polarity to the faces of specially-cut crystal slabs causes a change in shape which is used to drive a diaphragm or cone.

Rochelle salt crystals, which can be grown artificially from chemical solutions in quite large sizes and in a relatively short time, are most often

used in crystal loudspeakers. A typical Rochelle salt crystal slab might be cut to a size of about 2.5 inches square and $\frac{1}{8}$ inch thick, as shown in Fig. 9A. The slab must be cut from the crystal in a definite manner in order to secure a maximum change in shape when voltage is applied. When this crystal element is rigidly fastened at three of its corners, A. B and C, the application of an electric charge to its faces by means of tinfoil sheets cemented on each side will cause length A-D to increase or decrease, depending upon the manner in which the slab was originally cut from the complete crystal and the polarity of the applied charge. This change in length is of course quite small, but engineers have discovered how to utilize it to greatest advantage.

In the practical crystal loudspeaker, two slabs are cemented together. One increases in length when a voltage with a given polarity is applied, while the other decreases in length; the result is that the combined slab bends at its free corner through a distance far greater than the original change in length of either crystal. The principle is much like that of the bimetallic strip so widely used in some thermometers (two different metals are welded together in the form of a thin rectangular strip; one expands more than the other when temperature increases, with the result that the strip bends or curls with changes in temperature).

When two crystal elements are cemented together, giving a two-crystal unit, we have what is known as a *bi-morph cell*. The construction of the usual bi-morph cell is as shown in Fig. \mathcal{B} , where tinfoil sheets are cemented to each face of each crystal before the crystals are cemented together; in this way each crystal can be so charged that one will expand while the other contracts.

The construction of a typical crystal driving unit is shown in Fig. 9C. Three corners of the bi-morph crystal cell are rigidly supported between rubber pads, while a metal cap and a driving link or lever are cemented to the fourth corner. The entire cell is



FIG. 9. How does a crystal loudspeaker work? These three diagrams give you the complete story.

mounted in a water-proof housing, for Rochelle salt crystals dissolve in water and deteriorate in the presence of moisture. They are also affected by high temperatures and must not be used in locations hotter than 125° F. The motion of the free edge of the crystal unit is transferred to the diaphragm by the mechanical link or lever system.

Crystal driving units are essentially condensers; in fact, a 2.5 inch square bi-morph crystal cell has a capacity - of about .03 mfd. This means that its reactance will decrease at the higher frequencies, and it will therefore have a very good high-frequency response. Although crystal driving units can be designed to reproduce the entire audio frequency range, they are primarily used for the reproduction of the higher audio frequencies.

Loudspeakers Can Also Serve as Microphones

Before considering further details of sound-reproducing units, it is worth noting that any device for converting A.F. signals to sound can also be operated in reverse, so that sounds produced in the vicinity of the diaphragm or moving system will develop an electrical signal of corresponding wave form; in other words, a loudspeaker will also operate as a microphone. The better the fidelity of the loudspeaker as a sound reproducer, the better will be its performance as a microphone. One basic difference must be kept in mind; sound reproducers are designed to furnish large sound output powers, with the diaphragm or moving element pushing a large volume of air, while microphones are usually designed to respond to movements of small volumes of air or to weak sound inputs. In modern intercommunicating systems, small dynamic loudspeakers are widely used as microphones.

Action of a Horn as a Loudspeaker Coupling Unit

We have now considered the various methods ordinarily used for setting the diaphragm of a loudspeaker into vibration. Naturally we want this vibration of the diaphragm to produce a large sound output. Since the horn is a common air coupling system used for this purpose, it will be taken up next.

As you already know, the acoustical characteristics of a vibrating diaphragm in a driving unit may be represented by an electrical circuit like that shown in Fig. 10. The mass of the diaphragm is represented by L_m . (mechanical inductance), the compliance or springiness of the diaphragm is represented by C_m (mechanical capacitance), and the mechanical resistance of the diaphragm is represented by R_m . The mechanical force applied to the diaphragm by



FIG. 10. Equivalent mechanical circuit of a vibrating diaphragm in a loudspeaker.

the driving unit is represented by F_m ; for simplicity we can assume that it is an alternating force having a simple sine wave characteristic. We can then vary the frequency of this source and determine the effects of these variations on the performance of our loudspeaker.

The simple series circuit in Fig. 10 will serve as our guide for analyzing the action of a vibrating diaphragm. The applied force F_m causes the diaphragm to vibrate, and at any frequency the velocity of vibration will be governed by the mechanical reactances of L_m and C_m and by the mechanical resistance of R_m . When the two mechanical reactances are exactly equal, their effects cancel and we have mechanical resonance, with only $R_{\rm m}$ in the circuit to limit mechanical current flow. This mechanical current (i_m) represents the velocity of the diaphragm, and consequently also represents the velocity of air particles in the vicinity of the diaphragm. It is this air velocity which contributes toward the desired sound output. At input frequencies below the resonant frequency of the diaphragm, the mechanical reactance of $C_{\rm m}$ will be greater than that of $L_{\rm m}$, and the diaphragm will act as a mechanical capacitance in series with a mechanical resistance. At aboveresonance frequencies, the diaphragm will act as a mechanical inductance in series with a mechanical resistance but neither the mass (inductance) nor the compliance (capacity) involves loss of power; they merely offer opposition to changes in diaphragm position. It is the mechanical power absorbed by R_m which determines the loudspeaker sound output.

From Fig. 10 it is obvious that increasing the value of R_m increases the amount of mechanical power absorbed and therefore increases the sound output of a loudspeaker. Likewise increasing the driving force F_m will cause greater mechanical current flow, increasing the diaphragm displacement and velocity and thereby increasing the air velocity and the sound output power.

In the case of crystal or condenser driving units, the mechanical force $F_{\rm m}$ can be increased directly by increasing the A.F. voltage applied to the driving unit, while in other types of driving units an increase in the applied A.F. voltage will cause greater A.F. current to flow and this will increase the mechanical force $F_{\rm m}$. Increased mechanical force $F_{\rm m}$ obviously will cause greater diaphragm displacement and greater air velocity, with more power being absorbed by $R_{\rm m}$.

If a driving unit of a loudspeaker were placed in a vacuum, $R_{\rm m}$ would merely represent losses due to friction in the diaphragm itself; there would be no useful sound output since no air would be moved. The only limitations to the velocity of motion of the diaphragm would be this very low value of $R_{\rm m}$ and the difference between the mechanical reactances of $L_{\rm m}$ and $C_{\rm m}$; at resonance these mechanical reactances would balance out, and we could expect extremely large velocities and displacements of the diaphragm. Considering these facts from a practical standpoint, we can easily see that with a loudspeaker operating outside the gondola of a balloon high in the stratosphere (where atmospheric conditions approach a vacuum), the displacement of the diaphragm might be large enough to ruin the driving unit.

When the diaphragm is allowed to act upon air, a load is applied or coupled to the diaphragm and this is equivalent to increasing the value of $R_{\rm m}$. This is a desirable condition, for it allows us to secure a greater amount of useful work from the loudspeaker.

Measuring Loudspeaker Efficiency. When loudspeaker engineers desire to check the efficiency of a particular loudspeaker as a sound reproducer, they measure the electrical resistance of the loudspeaker input terminals under the following two conditions: 1, with the driving unit blocked, so that the diaphragm and the other moving elements cannot move; this gives the

resistance due to electrical losses in the driving unit only, and is known as the blocked resistance of the loudspeaker: 2, with the moving elements free to move." These resistance measurements are repeated for a number of frequencies over the entire audio range. The difference between the two measured resistance values at each audio frequency then represents the additional resistance due to the production of sound: this is called motional resistance. The ratio of the motional resistance to the total resistance when the diaphragm is free determines the loudspeaker efficiency; multiplying this ratio value by 100 gives the percentage efficiency. If, for example, these tests show that a loudspeaker has an efficiency of 10% at 1,000 cycles, the engineer knows that he will only get 2 watts of acoustical power out of the loudspeaker when he feeds 20 watts of electrical power into the loudspeaker.

Loading the Diaphragm. Loudspeaker engineers quickly learned that small diaphragms operating in free air without horns or other air coupling systems gave little sound output regardless of their velocity or displacement; measurements of loudspeaker efficiency verified this by indicating a very low motional resistance. The reason for this is easy to see; air directly in front of the diaphragm is alternately compressed and thinned out, and the same effect occurs at the back of the diaphragm. Whenever there is compression in front, there is thinning out or rarefaction at the back. Since compressed air tries to spread out or relay its effects to adjacent air particles in the easiest way possible, it merely moves around the diaphragm to the rear to equalize the thin air there, and only a very small volume of air is thus actually set into vibration by the diaphragm.





One way of overcoming this trouble is to use a large diaphragm or cone which will set a large volume of air into motion. Another way is to prevent the air in front of a small diaphragm from reacting on the air in back of the diaphragm. Either of these procedures increases the load on the diaphragm, thereby increasing the

^{*} In order to secure a true A.C. resistance measurement at the chosen frequency, the reactance of the voice coil is tuned out by means of a condenser in both cases.

motional resistance, producing more acoustical output and increasing the loudspeaker efficiency.

The mounting of a megaphone or conical-shaped horn-like that shown in Fig. 11A around the diaphragm of a driving unit was an early attempt to load the diaphragm by preventing air from escaping behind it, and also was intended to direct the moving air toward the listener. Results were not satisfactory, however, for the use of a conical-shaped horn did not take into account the fact that air in motion travels in definite natural paths. We have a small amount of air vi-



FIG. 11. Comparison of shapes of a conical horn with various types of exponential horns used with loudspeaker driving units.

brating at high velocity near the diaphragm; what we really want is a horn which will cause a greater amount of air a farther distance out to vibrate at a somewhat lower velocity, this process continuing until at the opening of the horn a very large volume of air will be vibrating at a low velocity. There must be no dead spots in the horn (regions unaffected by the vibration of the diaphragm); there must be no cancellation of effects along the length of the horn, no turbulence of air and no wasted energy. If all this is realized, the entire amount of air in the horn will be properly coupled to the driving unit, increasing the motional resistance and giving the desired increase in efficiency. Securing this desirable condition is known as matching the impedance of the air with the impedance of the driving unit; it serves to load the unit effectively and give a maximum of useful sound power output.

Exponential Horns. No doubt you are already familiar with loudspeaker horns. At the small end or throat of a horn the cross-sectional area is small, while at the large end or mouth the cross-sectional area is large and is a maximum. It makes little difference whether the cross-sectional shape of a horn is square, rectangular or circular; it is the manner in which the cross-sectional area increases from the throat to the mouth which determines the effectiveness of the coupling between the driving unit and the air in space. When this area increases according to what mathematicians call an exponential formula, giving a graceful curve like that shown by the dotted lines in Fig. 11A, the coupling will be most effective.

For a given throat area and a given mouth area, there can be any number of exponential horn shapes, but each would have different length and a different flare, as indicated in Fig. 11B. The flare of a horn refers to the <u>amount of spreading outward in a</u> given horn length: the shortest horn shown in Fig. 11B thus has the greatest flare. The amount of flare has a definite and important effect upon the frequency range of a horn loudspeaker, for it determines the efficiency of coupling at various frequencies in the audio range.

The formula used by loudspeaker design engineers in computing the shapes of various forms of exponential horns involves considerable mathematical knowledge, and will not be taken up here since it is seldom if ever needed by the practical man. There are certain facts to be derived from this formula which are of practical interest, however. The smaller the horns having smaller flares. Remember—a horn which is to provide effective coupling between the air and a loudspeaker driving unit at low frequencies should have a small flare.

Mouth Area. Although the area of the mouth of a horn does not have an



Assembly line in the Utah loudspeaker factory. Finished electrodynamic loudspeakers are placed on the conveyor belt which moves down the center of the long table.

the flare (amount of spreading out) of the horn, the lower will be the frequency which can effectively be coupled to the driving unit and the longer will be the horn for a definite throat and mouth area. That horn in Fig. 11B having the largest flare consequently cannot be expected to reproduce low frequencies as well as appreciable effect upon the cut-off frequency of the horn (the lowest audio frequency which it will radiate effectively), the mouth area does affect the uniformity of the loudspeaker response curve. A small mouth results in a sudden change in the velocity of air particles as they leave the horn, causing some of the sound waves to be reflected back into the horn. This gives rise to peaks and dips in the loading of the driving unit, causing a ragged sound output. In general, for a horn with a circular cross-section, the diameter of the mouth should be equal to at least onefourth the wavelength of the lowest frequency which is to be reproduced (the wavelength of sound is equal to its velocity of travel, 1,089 feet per Throat Area. The throat area of a horn is controlled essentially by the size of the diaphragm. It is common practice to design_driving_units for use with a circular throat having a diameter of 5% inch, in order that' standard horns and driving units can be used interchangeably. With a considerably larger throat area, a short horn with a small flare would provide a fairly low-frequency cut-off, but ex-



Four dynamic cone loudspeakers with single-piece spun aluminum exponential horns are here mounted on an ingenious carrying frame which can be used on any passenger car without damaging the roof. Rubber vacuum cups beneath each support prevent the assembly from sliding off while the car is in motion.

second, divided by its frequency in cycles; at 50 cycles, then, the wavelength would be $1,089 \div 50$, or about 22 feet). Thus a horn which is to handle frequencies down to 50 cycles should have a mouth diameter equal to at least $22 \div 4$, or 5.5 feet. The length of this horn would be quite long, for the small flare required to secure efficient coupling at this low audio frequency -would make necessary the long length in order to secure the required mouth area. perience has shown that such a horn would not give efficient coupling at higher frequencies. With a small diaphragm and the conventional small throat, good high-frequency output is possible but the movement or displacement of the diaphragm must be increased in order to secure good bass output. Yes, there are plenty of tough problems confronting the loudspeaker designer.

Construction of Horn. Any material is satisfactory for horn construction as long as it is dead as far as sound is concerned; in other words, the material must not vibrate under the influence of sound waves. Aluminum is an excellent material, but when used for the longer horns, it should be reinforced with ribs on the outside to prevent the large surfaces from vibrating in unison with the diaphragm. Plywood is widely used for horns of rectangular cross sections. Some horns are made of papier mache (molded paper fiber), while others are made of layers of cloth impregnated section which screws onto the driving unit is generally spun or cast from aluminum, and has the same form of exponential curve as the remaining sections of that particular horn.

Where space prevents the use of a long, straight horn having the desired characteristics (a horn length of 18 feet is by no means unusual where frequencies down to 50 cycles must be reproduced), the horn may be curled in the manner shown in Fig. 12B. The sharpest bend in the horn must not cancel any of the high-frequency



FIG. 12. Straight and curled exponential horns having standard 3/4" throats.

with a special glue or binder and formed to the desired shape. Horns which must 'withstand outdoor weather conditions generally are of sturdy metal construction; when plywood or some other non-metallic material is used, it should be treated to withstand moisture.

As a rule, a straight horn like that shown in Fig. 12A will give better results than a curled horn like that in Fig. 12B. It is common practice to build up a large horn in sections, different constructional procedures being followed for each type and size of horn and each type of material. That

sounds. If the difference in length between sound paths 3-4 and 1-2 in Fig. 12B, around a sharp bend, is less than one-half the wavelength of the highest frequency to be reproduced. the cancellation of sound will not be Cancellation occurs whenserious. ever two sound waves become 180° out of phase, and this can occur when one wave travels one-half wavelength more than the other. Curled horns can generally be designed to meet any reasonable space requirements while still observing the flare, mouth area and curvature specifications required for a desired performance.

Horn Ratings. If a horn is able to resist heavy sound pressures without self-vibration, it can handle unlimited sound output power. Good horns are therefore rated according to frequency range and not according to powerhandling ability. Either high- or lowpower driving units may be attached to any particular horn. Oftentimes several driving units are used on a single horn by redesigning the throat end of the horn or by inserting the proper type of connecting unit. In general, a long horn with a small flare and a large mouth will have a wide frequency range and will have high efficiency.

Design of Dynamic Driving Units for Horn Loudspeakers

In the high-power dynamic driving units used with horn loudspeakers, the



FIG. 13. Method of mounting horn with $\frac{1}{2}$ " throat on driving unit having a diaphragm considerably greater in diameter than $\frac{1}{2}$ ", and equivalent mechanical circuit for such a combination.

diaphragm is generally made considerably larger than the 5% inch throat diameter of the norn, in order to move a larger volume of air and secure a higher initial air velocity. The arrangement of a dynamic driving unit might be as shown in Fig. $13A_{...}$ The diaphragm is made in the shape of a cone, to prevent buckling when a



Marine-type mounting of dynamic cone loudspeaker. This weather-proof unit, made by Atlas Sound Corp., gives excellent distribution of sound and at the same time is capable of withstanding the extremely severe weather conditions encountered on board ships, on sound trucks and even on fire trucks. Tests have shown that a stream of water can be played directly into the mouth of the loudspeaker without affecting loudspeaker operation. The horn is made from heavy-gauge aluminum and steel spinnings. Arrow lines show paths of sound waves.

strong force is applied at its center by the voice coil.

The mounting of the diaphragm is such that it has a certain amount of natural springiness, which returns it to a normal position when no current flows through the voice coil, or the cone and the chamber in back of it are made air-tight so that the air space in back of the diaphragm will act as a cushion or spring which returns the diaphragm to its normal position. The diaphragm therefore has mechanical capacitance. Furthermore, since the diaphragm and voice coil together have a certain amount of mass, they also have mechanical inductance. The experience of loudspeaker engineers has shown that these act in series, as shown in Fig. 13B. The load which is placed upon the cone-shaped diaphragm by the exponential horn can be represented as • a mechanical resistance acting in series with the mechanical inductance and mechanical capacitance. We must not overlook the air chamber between the cone and the surface directly in front of it which supports the throat end of the horn. This air the compliance of the rear chamber (C_B) form a series resonant circuit. Resonance will occur at some particular frequency, and at resonance the input force F_m will depend upon the length of the wire in the voice coil, upon the voice coil current and upon the flux density in the air gap surrounding the voice coil. Furthermore, at resonance the input force will be acting solely upon mechanical resistance R, and practically all of the in-



Courtesy Racon Electric Co., Inc.

Battery of trumpet horn loudspeakers set up in front of the Hotel Astor at Times Square, New York City for a special event. This arrangement for radiating sound equally well in all directions requires a number of horns, with separate driving units for each.

space can be considered an extra mechanical capacitance $C_{\mathbf{F}}$ acting in parallel with the mechanical resistance.

We know how an electrical circuit resembling that in Fig. 13B will behave under various conditions, since it is a common radio circuit, and consequently we can predict the behavior of our loudspeaker by studying this circuit.

Assuming that the value of $C_{\mathbf{F}}$ is small, as it ordinarily is, we can see that the mass of the cone (L) and put force will serve to produce useful sounds (assuming a perfect exponential horn). The loudspeaker designer can select the values of L and C_B so that mechanical resonance occurs at a low audio frequency, thus providing reinforcement of bass notes, or at some intermediate frequency. This is one way of securing a desired frequency response for a loudspeaker.

At high frequencies the mechanical reactance of $C_{\rm F}$ becomes important Referring to our equivalent mechanical circuit in Fig. 13B, $C_{\rm F}$ by-passes high-frequency signals around the load, and consequently in the loudspeaker this air chamber ahead of the cone definitely suppresses the high audio frequencies. The volume of air in this chamber determines to a considerable extent its mechanical capacity, and therefore the effects of this chamber can be reduced by making it as small as possible while still allowing ample room for cone movement. The shape of this air chamber



account when designing the driving unit for a horn loudspeaker.

A number of typical mounts for the throat and diaphragm of a dynamic horn loudspeaker are shown in Fig. 14. In each case the air chamber ahead of the diaphragm or cone is designed to give desired performance while reducing undesirable effects. When a wide range of sound frequencies is to be reproduced, it is generally best to use two loudspeak-



FIG. 14. Constructional features of three typical dynamic driving units are shown here. A: Unit employing a ring-shaped air chamber between the cone and the throat (secured by mounting a plug in the throat). The chamber is carefully shaped to reduce cancellation of sound waves at the higher frequencies. B: Unit using multiple outlet paths from diaphragm to throat (secured by forming air holes in a plug) to release the back pressure. C: Complete dynamic driving unit energizing a modification of the ring-shaped air chamber shown at A.

alongside the throat is important for another reason; if the shape is such that air in the chamber can take several paths, of varying length, to the throat of the horn, it is possible that some of the high-frequency sounds will reach the throat out of phase and will cancel. In high-power horns the air pressure in this chamber may be tremendous; air cannot be compressed uniformly under these conditions, and distorted wave forms are the result. All these factors must be taken into ers; one loudspeaker for frequencies from 50 cycles to 6,000 cycles, called a "woofer," and another loudspeaker for frequencies from 6,000 cycles to 15,000 cycles, called a "tweeter."

The science of horn loudspeaker design has now advanced to the point where efficiencies of up to 50% can be expected, with reasonably flat response. This means that the best horn loudspeakers will deliver 5 watts of sound power for each 10 watts of electrical input to the voice coil.

World Radio History

TEST OUESTIONS

Be sure to number your Answer Sheet 26FR-1.

Place your Student Number on every Answer Sheet.

Never hold up one set of lesson answers until you have another ready to send in. Send each lesson in by itself before you start on the next lesson. In this way we will be able to work together much more closely, you'll get more out of your Course, and you will receive the best possible lesson service.

- 1. Name the three sections of a complete loudspeaker.
- briving unit, Cone of air Counter; 2. Into what four general groups can loudspeakers be divided?
- 3. How is the underrable frequency-doubling effect eliminated in a bi-polar headphone unit? By use of a strong Verme.
- 4. What does mass in a mechanical vibrating system correspond to in an electrical oscillating circuit? Induchance of a coul
- 5. What determines the natural frequency of vibration of any physical object? mars & compliance mers
- 6. What three things determine the force acting upon the voice coil of a dynamic loudspeaker? P. 10
- 7. What is the basic action of a crystal loudspeaker? 8. What is meant by the flare of a horn? and crystal registican
- 9. If a horn is to provide effective coupling between the air and a loudspeaker driving unit at low frequencies, should a large or a small flare be used? The smalle, the place the lower the

, clabs Care

10. Are horns rated according to power-handling ability? No. I'my an rated according to free. Range.
AT THE END OF THE RAINBOW

The only pot of gold you'll find at the end of the rainbow is the one which you put there yourself.

Now, when your best earning years are still ahead, is the time for you to fill that pot of gold. You're an N.R.I. student—you're carrying the ball down the field right now for a touchdown—and everything favors you to make the goal you have in mind.

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There is no royal road to anything. Steady progress step by step will get you anywhere and bring you a success which endures. The only failure you need to fear now is failure to stick to the goal you know is best.

J. E. SMITH



STUDY SCHEDULE No. 27

For each study step, read the assigned pages first at your usual speed, then reread slowly one or more times. Finish with one quick reading to fix the important facts firmly in your mind, then answer the Lesson Questions specified for that step. Study each other_ step in this same way.

1. Cone Loudspeakers; Dynamic Loudspeaker Design

Problems - - - - - - - - - - - - - - - - - Pages 1-7

Shape and Size of Cones; Cone Mountings; Cones of Dynamic Loudspeakers; "Tweeter" Cones; The Spider; Mechanical Resonance in Dynamic Loudspeakers; Piston Action; Double Voice Coils. Answer Lesson Questions 1, 2, 3 and 4.

2. Baffles; Box Baffles - - - - - - - - - - - Pages 7-16

Purpose of a Baffle; Irregular-Shaped Baffles; Mechanical Characteristics of Box Baffles; Equivalent Electrical Circuit of a Loudspeaker-Baffle Assembly; Natural Resonant Frequency; Infinite Baffles; Standing Waves; Bass-Reinforcing Box Baffles; Stromberg-Carlson Acoustical Labyrinth; Special RCA Box Baffles; Acoustic Clarifiers. Answer Lesson Questions 5, 6, 7 and 8.

🗍 3. Sound Diffusers; Horn-Shaped Baffles - - - - - - Pages 16-19

Polar Radiation Patterns; Deflecting Vanes; Tweeter and Woofer Loudspeakers; Types of Horn-Shaped Baffles; Loudspeaker Efficiencies; Exponential Horns. Answer Lesson Question 9.

☐ 4. Loudspeaker Impedance - - - - - - - - - Pages 19-24

Mechanical Inductance, Capacitance and Resistance of a Loudspeaker; Electrical Impedance of Loudspeaker; Determining Loudspeaker Impedance; Matching, Typical Loudspeaker Matching Circuits; Replacement of Output Transformers. Give special attention to the last three subjects here, because they are important practical applications of the theoretical information in the first part of this study step.

5. Field Coil Connections; Acoustical Problems of High-

Fidelity Reproduction - - - - - - - - - - Pages 24-28

Use of Field Coil as Filter Choke; Types of Field Coil Connections; Separate Field Supplies; Elimination of Hum; Shading Ring; Hum-Bucking Coil; High-Fidelity Two-Loudspeaker Systems and their Response Curves; Room Acoustics; Loudspeaker Replacement Hints. Answer Lesson Question 10.

☐ 6. Mail Your Answers for this Lesson to N. R. I. for Grading.

7. Start Studying the Next Lesson.

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HOW CONE-TYPE LOUDSPEAKERS WORK

Cone Loudspeakers

LOUDSPEAKERS of the horn type, having the required wide frequency range for high-fidelity reproduction, are too bulky for use with the average home radio receiver even when curled to occupy a minimum of space. Engineers recognized this fact soon after loudspeakers first came into use, and eliminated the horn entirely by increasing the size of the diaphragm and setting a large amount of air directly into motion.

Since a large, flat diaphragm will buckle or bend readily even when driven from its center, loudspeaker engineers shaped this diaphragm in the form of a cone, attaching the driving unit to its apex or point. The outer edges of the cone are free to move, being held in alignment by a soft leather ring or washer cemented to the edges of the cone and supported by a circular steel frame. The entire cone moves in and out in unison with the driving unit, at least for low frequencies.

In the early days of radio, a cone as large as 36 inches in diameter was quite common, being used to secure good low-frequency output, but the high-frequency response of these cones was irregular. Today cones are generally from 6 to 14 inches in diameter; because loudspeaker engineers now have a better understanding of the exact operation of a cone loudspeaker, remarkably flat response characteristics are being secured.

The cone arrangement of a typical cone loudspeaker is shown in Fig. 1. Observe that the frame of the cone is held against a flat surface known as a baffle, which may be either of wood, plywood, fiber board or other soundabsorbing material in which a circular hole the same diameter as the cone has been cut. This baffle board serves to prevent sound waves produced at the rear of the cone from interfering with sounds produced at the front of the cone. For the present we will assume that the baffle is so large that no interference or can-



FIG. 1. Cross-section view of a cone loudspeaker and its supporting structure, as mounted on a wood baffle. Drlving unit is not shown.

cellation of sound can take place; later in this text-book we will consider baffle design in detail.

In general, cone loudspeakers will be found with balanced armature magnetic driving units, with electrodynamic or permanent magnet dynamic driving units, and with crystal driving units. A number of typical cone loudspeakers are shown in Fig. 2.

The unit in Fig. 2A is typical of cone loudspeakers using a balanced armature magnetic driving unit. The apex or point of the cone is driven by a metal extension arm or lever attached to the vibrating armature. In many cone loudspeakers, the mechanical lever system is so arranged that it provides mechanical amplification of the cone movement; in other words, the apex of the cone may move twice or three times as far as the point on the armature to which the lever is attached. Even with this mechanical lever arrangement, the movement of the cone is definitely limited by the movement of the balanced armature between the pole pieces. Magnetic loudspeaker units of this type ordinarily will not handle much more than 2 watts of electrical input power. sume this to be an electrodynamic unit, with the permanent magnetic flux being produced by direct current flowing through the field coil. If there are no additional leads, the unit is of the permanent magnet dynamic type.

The cone unit alone of a dynamic loudspeaker is shown in Fig. 2D. Observe that the voice coil is attached directly to the cone. The size of the voice coil leads has been exaggerated



FIG. 2. Typical cone loudspeaker units.

In Figs. 2B and 2C are two views of a representative dynamic cone loudspeaker. The transformer which matches the voice coil impedance to the output impedance of the power stage of the receiver can be seen mounted directly on the loudspeaker frame, as is the usual practice. If, in addition to the leads coming from the matching transformer, there are two or three more leads coming from the pot or cylindrical housing at the back of the loudspeaker, we can asin the drawing in order to make them visible; actually they are made of very fine flexible wire. The outside edges of the cone are cemented to a cardboard ring, with corrugations or a soft leather ring just inside the cardboard to permit free movement of the entire cone. The cardboard ring is in turn either cemented or bolted to the steel ring which is a part of the loudspeaker frame. This type of cone construction is widely used in order to simplify replacement

World Radio History

of cones which have become damaged or have deteriorated through continuous use.

A permanent magnet dynamic cone loudspeaker is shown in Fig. 2E. Here the large U-shaped steel bar serves as the permanent magnet. Both tips of this bar are magnetized with the same polarity, the center being of opposite polarity. The soft iron central core for the voice coil is attached to the center of the permanent magnet, and is therefore opposite in polarity to the soft iron pole piece which is in contact with the cnds of the U-shaped permanent magnet.

A drystal cone loudspeaker unit is shown in Fig. 2F. Ordinarily this will be found with a small, light cone about 5 inches in diameter. Crystal loudspeakers are often called "tweeters," and are used to reproduce the higher audio frequencies, in conjunction with an ordinary loudspeaker designed to reproduce the bass notes.

Dynamic Loudspeaker Design Problems

The Spider. One important part of a dynamic cone loudspeaker has not yet been taken up, simply because it does not appear in ordinary loudspeaker illustrations; this part is the spider, a springy sheet of fiber or bakelite material cut as shown in Figs. 3A and 3B. The spider serves two purposes, that of <u>centering the</u> voice coil with respect to the soft iron core and pole pieces, and that of <u>returning the voice coil and cone to a</u> <u>normal position</u> when the driving force drops to zero.

In horn loudspeakers the back of the driving unit is completely enclosed and the air in the enclosed rear chamber provides the springiness required to return the diaphragm to its normal position; in dynamic cone loudspeakers, however, this back air chamber is usually absent. The ordinary cone has little natural springiness because of the nature of its mounting, and therefore a spring must be used to provide the restoring action which is essential to correct loudspeaker operation. This spring is called a *spider*.

The *internal* spider unit shown in Fig. 3A is cemented inside the voice coil at the point where it joins the cone. The center part of the spider is fastened to the cylindrical iron core



FIG. 3. Typical spiders used in cone dynamic loudspeakers to keep the cone and voice coil assembly properly centered with respect to the central iron core. That at A is for the front of the cone, and the form shown at B is for mounting on the back of the cone, around the voice coil. Spiders also provide a restoring spring action for the cone. Spiders of the type shown at B, which must surround the voice coil, will be considerably larger than those at A, which fit inside the voice coil.

inside the voice coil but is held away from the core by a bushing. A machine screw passing through the center hole of the spider and the bushing firmly anchors the spider. When this screw is loosened, the spider, voice coil and cone assembly can be moved a small amount in any direction to permit exact centering of the voice coil.

Sometimes the spider is of the shape shown in Fig. *SB*, and is cemented to the outside of the voice coil at the point where it joins the cone. Three machine screws, one for each leg of the spider, hold it firmly against the outside housing or pot of the loudspeaker. Occasionally you will find an *external* spider of this type made from webbed cloth which has been treated with bakelite varnish to give it springiness.

Since dynamic loudspeakers are by far the most common of any types in use today, it will be worth while to consider in detail a number of their peculiar design features which are not apparent at first sight. A great many design problems must be solved in connection with the driving unit, the cone and the spider in order to secure good frequency response and high efficiency.

You already know that the force acting upon the cone of a dynamic loudspeaker is determined by the current flowing through the voice coil, by the length of the wire used in the voice coil, and by the flux density in the vicinity of the voice coil. Before we can determine the effectiveness of this force in producing air velocity or sound, however, we must analyze the characteristics of our moving or vibrating system. Referring to Fig. 4A, we can see that the voice coil has a certain amount of mass, which we can consider as mechanical inductance $L_{\mathbf{v}}$. The cone has a mass which can be considered as mechanical inductance $L_{\rm C}$, and the air which is moved directly by the cone also has a certain amount of mass, which can be represented as mechanical inductance L_{A} . The spider provides most of the springiness or compliance, and this can be represented as mechanical capacitance C_8 . The opposition offered by air particles to the movement of the diaphragm can likewise be represented as mechanical resistance R_{A} . Experiments have shown that all these parts can be considered as acting in series, as shown in Fig. 4B. Note the absence of a mechanical capacitance shunting R_{A} ; there being no air pocket in front of the cone, the dynamic loudspeaker does

not have this capacitance which in a horn unit reduces the high audio frequency output. The grille cloth usually found in front of a dynamic loudspeaker is loosely woven, so that sound waves can travel through it readily at all audio frequencies.

Mechanical Resonance in Dynamic Loudspeakers. From our knowledge of electrical circuits like that shown in Fig. 4B, we know that mechanical resonance will occur in our vibrating system when the combined mechani-



Courtesy Utah Radio Products Co.

The external spider, surrounding the cone end of the voice coil, can be seen on this photograph of a Utah dynamic loudspeaker.

cal reactances of $L_{\mathbf{v}}$, $L_{\mathbf{c}}$, and $L_{\mathbf{A}}$ are exactly equal to the reactance of C_8 . In a dynamic loudspeaker the spider (or whatever mechanical restoring spring action is used) has a high compliance, and hence the mechanical capacity is high. High values of capacity acting with high values of mechanical inductance make mechanical resonance occur at low frequencies in dynamic cone loudspeakers. In the average dynamic cone loudspeaker unit, resonance occurs below 100 cycles. This resonant action is usually utilized to reinforce the response of the loudspeaker at bass frequencies.

At resonance the entire input power is utilized in producing useful sounds; unless the cone is properly loaded by means of a suitable baffle, the movements of the cone will be excessive at resonance, resulting in distortion or even in damage to the voice coil, and particularly to its spider.

Piston Action. At frequencies above resonance, the circuit acts as a mechanical inductance in series with the mechanical resistance, which means that the driving force has only the unit decreases with increasing frequency. Although this would tend to give greater output at high frequencies, unfortunately the cone will vibrate as indicated in Fig. 4C only for certain frequencies. This means that the high-frequency response of the loudspeaker will have many irregular peaks rather than the desired uniform response.

In a 12-inch cone (the diameter of the free edge), this change from piston action to vibrating cone action



FIG. 4. A cone loudspeaker like that at A can be studied by means of an equivalent acoustical circuit like that at B. The diagram at C illustrates how a cone itself can vibrate at high audio frequencies; the points marked N represent nodes, which are points at which there is no vibration. The dotted line curve represents the maximum amplitude of the vibration at any other point along the cone.

mass and resistance to oppose it. You know that inductive reactance increases with frequency, and consequently our mass reactance in this mechanical system would likewise increase if it were not for the fact that the mechanical resistance R_{A} increases also and compensates for the increased mass to a certain extent. (Actually, tests have shown that the air resistance R, increases uniformly with frequency up to the point where the diameter of the cone becomes onehalf wavelength.) This essentially flat or uniform output response is obtained up to the frequency where the cone begins to vibrate with nodes and peaks as shown in Fig 4C. Under this condition the cone is no longer acting upon the air in the manner of a piston, but is itself vibrating. Once this condition is reached, the mechanical resistance R_{A} stays essentially constant, while the mass reactance of takes place at about 750 cycles. For an 8-inch cone the change occurs at about 1,000 cycles, and for smallerdiameter cones it occurs at correspondingly higher frequencies.

Since pure piston action is desirable in the cone of a dynamic loudspeaker in order to secure a flatter frequencyresponse, the loudspeaker designer uses a scheme which automatically reduces the diameter of the cone for the higher frequencies. One such scheme involves placing a number of concentric corrugations in the cone, as shown in Fig. 5A; the result is that at low frequencies the entire cone will be effective, with edge 1 serving as the free edge. At slightly higher frequencies the free edge of the cone will move inward to the corrugation at point 2; under this condition only that part of the cone between corrugation 2 and the voice coil will be in motion. At increasingly higher

frequencies, the effective cone diameter becomes increasingly less; at the highest frequency, corrugation \mathcal{S} becomes the free edge.

A typical dynamic cone loudspeaker having corrugations in the



FIG. 5. Two methods of improving the high frequeocy response of a cone loudspeaker are illustrated here.

cone to reduce self-vibration and thereby improve the high-frequency response is shown in Fig. 6.

Another scheme for increasing efficiency by securing pure piston action involves using a special shape of cone, the free edge of which will automatically shift towards the voice coil as the audio frequency increases. A suitable shape for accomplishing this result is shown in Fig. δB ; since the curve of this cone corresponds to that of a parabola (a special geometric curve), it is often called a *para-curve* or a *curvilinear* diaphragm.

A third scheme involves gradual thinning out of the cone material from the voice coil toward the free edge. Each of these schemes gives a reasonably flat frequency response up to about 3,000 cycles, with decreased and somewhat non-uniform output at higher frequencies. This does not mean, however, that no high-frequency output is obtained; actually the highfrequency output can be quite satisfactory for ordinary radio receiver requirements, but for more nearly perfect reproduction, a second loudspeaker, designed specifically for uniform high-frequency reproduction, should be considered.

Double Voice Coils. Even if we could limit cone movement to that region immediately in the vicinity of the voice coil at high frequencies, in order to secure the desired piston action, the voice coil itself would still have too much mass for perfect results. This mass can-be reduced by careful design of the voice coil in order to keep its weight at a minimum, and this in turn extends the upper frequency range of the loud-



FIG. 6. The corrugations which serve to reduce the effective diameter of the cone at higher frequencies are clearly visible in this photograph of an RCA Victor dynamic cone loudspeaker unit.

speaker certain a amount. To further extend this range, two voice coils are sometimes used together, as illustrated in Fig. 7. The coils are connected by an elastic coupling band which can be represented as C_1 . Coil L_1 has considerably greater mass than L_2 . At low frequencies the mechanical capacitance C_1 is ineffective (the elastic coupling does not bend), and the driving forces produced by both coils act upon the cone. At high frequencies only the smaller coil is capable of moving at the high rate of vibration involved, and the elastic coupling at C_1 allows this coil alone to drive the cone, while L_1 , remains

essentially fixed. In this way the cone can be made to produce uniform outputs up to about 6,000 cycles, with lowered output at higher frequencies than this. Coils L_1 and L_2 each have their own leads, being externally con-



FIG. 7. The double voice coil construction shown here is sometimes used to reduce the effective mass of the vibrating system at high frequencies and thus improve the high-frequency response-

nected together in series. L_1 , the heavier coil, is ordinarily shunted with a condenser whose value is such that high-frequency currents will be by-passed around it and will flow only through L_2 , where they will produce a useful mechanical force.

Baffles

When the voice coil pushes the cone of a dynamic loudspeaker forward or away from the pot, the air in front of the cone is compressed or made heavy, while the air at the rear of the cone is simultaneously thinned out or made rare, as indicated in Fig. 8A. Those particles of air which are compressed will exert force on adjacent air particles, transferring this compressed condition to nearby air particles at the speed of sound, which is about 1,089 feet per second. It is a natural tendency for these compressed air particles to move outward in all directions, with many of them moving around the edges of the cone to the rear. If the air at the rear of the cone is rarefied at the time when compressed air particles arrive, the particles rush to the rear to equalize the air pressure, and this rush of particles in turn brings more compressed air particles from the front to the rear. The result is cancellation of useful sound, since sound is produced the instant that air is compressed at the front of the cone.

To limit this natural tendency for air to avoid doing useful work, we can place around the loudspeaker a baffle made up of a square or circular board having cut in it a hole equal to the diameter of the cone. A baffle such as this is indicated in Fig. BB; we can immediately see that air particles which are compressed in front of the cone must travel completely around the baffle before they can reach the back of the cone. It takes a certain amount of time for this compressed air condition to be transferred from



FIG. 8. The shortest sound wave path from the front of a cone around the baffle to the rear must be at least one-half the wavelength of the lowest sound frequency to be reproduced if cancellation of sound waves is to be avoided.

point 1 around the baffle to point 2, and this time should be made long enough to allow the cone to begin compressing the air at the rear on its backward movement. Under this condition there is practically no cancellation of compressed air by rarefied air, and a considerably greater amount of useful sound is radiated by the front of the loudspeaker.

Let us assume that the cone is moving in and out in a sine wave manner. When the cone is farthest forward, compression of air at point 1 will be a maximum. This compressed air will take the shortest path to point 2, where a rarefied condition exists at this moment; naturally this will be path L around the baffle. The time taken for air movement along this path will be equal to the length of the path divided by the speed of sound; if this time is such that the cone has gone through a complete half cycle by the time the compressed air reaches point \mathcal{Z} , then the cone will be compressing air at point \mathcal{Z} , the compressed air from that point will meet air which is equally compressed, and no cancellation will take place. As a result, the compressed air at the front will be forced to move in the desired



FIG. 9. The irregular-shaped flat baffle at C gives more uniform frequency response than either the circular baffle at A or the square baffle at B. Dotted lines indicate possible sound wave paths.

forward direction. The path length required to prevent front-to-rear cancellation can readily be figured out by dividing 1,089 by twice the frequency of the lowest sound signal to be reproduced. For example, if the lowest frequency to be reproduced is 50 cycles, the minimum path length Lwould be equal to 1,089 divided by 2 times 50, which is 10.89 feet.

The longer the path L around the baffle, the lower will be the cut-off frequency of the loudspeaker. Where space is limited, as it is in the conventional radio receiver cabinet, the loudspeaker is usually mounted in a box like that shown in Fig. \mathcal{SC} . Here path L would be measured as indicated.

Although each baffle has a definite cut-off frequency, with sounds below this frequency being eliminated or greatly reduced in strength, this is no assurance that frequencies above the cut-off value will be reproduced without attenuation. Suppose that the diameter of the circular baffle shown in Fig. 9A is such that cut-off occurs at 100 cycles. When a 200-cycle sound is fed to the loudspeaker mounted on this baffle, the cone will go through one complete cycle during the time required for the sound to travel from the front to the rear. and consequently the compressed air traveling around the baffle will encounter rarefied air at the rear. Cancellation takes place and loudspeaker output is low at 200 cycles. This same effect will occur at each higher multiple of the cut-off frequency (3) times, 4 times, 5 times, etc.), but in the case of these higher multiples the cone will have gone through so many cycles during the time required for the compressed air to get around the baffle that most of the sound produced at the front of the cone will have traveled too far out to be affected. It is at the frequency equal to twice the cut-off frequency that cancellation is the most serious.

Irregular-Shaped Baffles. With a square baffle like that shown in Fig. 9B, the paths around the baffle are of many different lengths, so that even though cancellation occurs on some paths there will always be other paths at which a particular frequency is not entirely cancelled out. The result is that the frequency response of the loudspeaker system is far more uniform in the low frequency region down to the minimum cut-off frequency (determined by the longest baffle path length). The most uniform response at low frequencies is obtained with an extremely large baffle or with a special form of completely enclosed box baffle, sometimes

known as an *infinite baffle*. An irregular shape of baffle like that shown in Fig. 9C is better than a square or circular baffle, for this provides a number of widely different path lengths.

Box Baffles

The cabinet in which the cone loudspeaker of the average home radio receiver is mounted is essentially a box. Although this box does give the desired long path for sound around the baffle, the presence of the box or cavity behind the loudspeaker is not always desirable. This cavity can in itself be a sound resonator, acting much in the manner of a horn and setting air into vibration at some frequency. In the average radio receiver cabinet this cavity can be shocked into vibration at a low or bass frequency in many cases, producing an undesirable high bass output. The cavity is in addition a load on the cone, increasing its mechanical resistance and altering the frequency response of the loudspeaker; on the other hand, if this load produced by the cavity is properly proportioned, it can be a real aid to loudspeaker operation. Since many loudspeaker installations take into account the design of the back resonating chamber or box baffle, we will analyze this problem in greater detail in order that we can better understand the unique constructions which are sometimes employed.

With a cone mounted on a large flat baffle, the back of the cone must move equally as large a mass of air as the front of the cone. This mass has an inertia which opposes any force which tries to set it into motion, and consequently we can consider this mass as a mechanical inductance.

With a flat baffle, the mass of air in back of the cone has little stiffness. for it can expand equally well in all directions behind the baffle. As we bend back the edges of the baffle to form a box-like chamber around the rear of the cone, we confine the mass of air to a limited region and reduce the mass upon which the back of the cone is compelled to act. Confining this air to a definite volume increases its stiffness, with the result that



Typical electrodynamic loudspeaker, available from Jensen Radio Mfg. Co., Chicago, Ill., in 5, 6 and 8-inch sizes (cone diameters) with various output transformers and various field coil resistance values for replacement purposes.

greater force must be applied to the cone in order to move this air. This reduces the compliance of the rear chamber, thereby reducing its mechanical capacitance. (In the case of a flat baffle, this mechanical capacitance was so high that its effect could be neglected.) The mechanical force must therefore act simultaneously upon a mechanical capacitance and a mechanical inductance in parallel.

We know also that a certain amount of sound always escapes from the rear of the chamber; this radiated sound must be represented by mechanical resistance acting in series with the air mass. A completely closed box has no sound radiation at the rear and therefore has zero mechanical resistance. A flat baffle has a large mechanical resistance, its value being equal to the mechanical resistance existing at the front of the cone. An open box has a medium value of mechanical resistance.

With these facts in mind, we can set up the equivalent mechanical circuit of a loudspeaker mounted in a box baffle, and use this as our guide in analyzing the good and bad fea-



FIG. 10. Rear chamber of a box baffle, and equivalent circuit of the entire system, including the chamber.

tures of this particular baffle. Figure 10A shows a cone and voice coil mounted in a box baffle, and Fig. 10B represents the equivalent circuit including the back chamber. The mechanical force acting upon the entire system varies in an alternating manner corresponding to the variation in voice coil current. This force acts first upon the mass of the cone and voice coil assembly, represented in Fig. 10B as mechanical inductance $L_{\rm C}$, upon the spider which provides springiness or mechanical capacitance $C_{\rm c}$, and upon the mechanical resistance due to sound radiation from the front of the cone, represented as $R_{\rm C}$. The additional load placed upon the cone by the back chamber can be represented by mechanical capacitance $C_{\mathbf{B}}$ in parallel with mechanical inductance $L_{\rm B}$, representing the mass

of the back chamber air; the mechanical resistance introduced by sound radiated from the rear of the baffle is represented as $R_{\rm B}$, acting in series with $L_{\rm B}$.

We can now see that $C_{\mathbf{B}}$ and $L_{\mathbf{B}}$ in Fig. 10B together form a parallel resonant circuit. Furthermore, this resonant circuit can be shocked into selfoscillation by the driving force Facting through the reactances of $L_{\rm C}$ and $C_{\mathbf{c}}$, under which condition the air in the cavity will vibrate and radiate sound through the cone. This self-produced sound may set the cone into vibration, producing frequencies which were not present in the original program and giving the so-called boomy response which some loudspeakers have. Furthermore, if the driving force F has the same frequency as the resonant frequency of the cone loudspeaker unit (resonant circuit $L_{\rm C}$ - $C_{\rm C}$). the cone will be set into excessive vibration, giving a whooping sound. This occurs because $L_{\rm C}$ - $C_{\rm C}$ acts as a low mechanical resistance at resonance, allowing more of the driving force F to be applied to the cavity and increasing the sound output of the back chamber at that particular frequency.

Natural Resonant Frequency. Increasing the size of a box baffle increases the volume of the cavity and thereby increases the values of mechanical capacitance $C_{\mathbf{B}}$ and mechanical inductance $L_{\mathbf{B}}$; this lowers the natural resonant frequency of the cavity (the resonant frequency of the parallel resonant circuit). If the box is made sufficiently large or if an infinite flat baffle (where no sound waves whatsoever can escape to the rear of the cone around the baffle) is used, the natural frequency of the region behind the cone can be made so low that the effects of cavity resonance will not be heard at all.

A radio receiver which has an open back is generally designed so that when the cabinet is kept a few inches away from the wall or is placed in a corner so there is free movement of air behind the cabinet, $C_{\rm B}$ will be high enough in value to prevent audible cavity resonance effects. Placing the radio receiver back up against the wall tends to close up the back chamber and reduce $C_{\mathbf{B}}$; under this condition cavity resonance may occur at an audible bass frequency, in the form of boominess and the whooping sounds mentioned before. Remember: radio receivers with open backs will sound best when kept a few inches away from the wall or when located across a corner of a room.

Infinite Baffles. Suppose that we closed up the back of the cabinet com-



FIG. 11. These curves show how cone velocity (which determines sound output) varies with frequency for an ordinary box baffle (curve 1) and for a box baffle which is tuned to the natural resonant frequency of the moving system (curve 2).

pletely, thereby making the values of $C_{\rm B}$ and $L_{\rm B}$ constant. This, of course, does not in itself eliminate cavity resonance, but if we make the resonant frequency of $C_{\rm B}$ - $L_{\rm B}$ equal to the resonant frequency of $L_{\rm O}$ - $C_{\rm C}$, an entirely new effect will take place. When the driving force F has this same frequency, the series resonant circuit will act as a very small resistance, but the parallel resonant circuit will act as a very high resistance, greatly limiting the flow of mechanical current *i*. This means that there

will be no vibration of the cone at resonance, no whooping sounds, no sound waves will be radiated by the back of the cabinet, and we will have what is known as an *infinite baffle*. Figure 11 shows more clearly what happens under this condition; curve 1 shows how cone velocity, which determines sound output to a great



Courtesy Oxford-Tartak Radio Corp.



extent, increases gradually with frequency up to a maximum value at a frequency equal to the natural resonant frequency of the cavity. At frequencies above cavity resonance, the cone velocity drops again. Curve 2 is for the condition where the cavity is tuned to the natural resonant frequency of the cone; you can see that the peak has been reduced considerably at resonance, while cone velocity is still high enough to give the system desired bass reinforcement. Adjusting the cavity resonant frequency to the natural frequency of the cone system will greatly reduce the undesirable effects of cavity resonance in a box type baffle.

Standing Waves. At frequencies higher than the cavity-resonance frequency, the cavity will act as a lowreactance condenser which does not seriously affect the response of the loudspeaker. At the higher sound frequencies, however, where the wavelengths involved become shorter than the dimensions of the box, there will be considerable reflection of sound from one side of the box to the other. If these reflections occur continually over the same path, the compressions and rarefactions of air will be reinforced at definite points. and standing sound waves will occur; these give resonant effects much like those produced in organ pipes and in musical wind instruments. In other words, sounds at certain frequencies will be prolonged somewhat, due to the production of harmonics, and the reproduced sound will not be natural.

Using a shallow cavity or making it irregular in shape so that reflected sounds cannot build up along the same path will cure this trouble. Other remedies involve lining the box with a sound-absorbing material, hanging heavy sound-absorbing materials down from the top inside the box, or lining the sides of the box with rectangular pieces of wood, thereby preventing sounds from being reflected back and forth over the same path. Whenever sound waves are reflected repeatedly over a path which is some multiple of a half wave in length, compression will occur at the same points along this path for each reflection. Reinforcement of sound occurs under this condition due to the building up of pressure, and the resulting standing waves produce annoying sounds. The box itself should be made of heavy wood, so it will not vibrate at the lower bass frequencies.

So far we have considered box baffle arrangements which are intended to prevent cavity resonance at low frequencies and standing waves at high frequencies. In each case the sounds produced by the rear of the cone were either suppressed or absorbed, and were therefore wasted. Bass-Reinforcing Box Baffles. Increased bass output is practically always desirable in a loudspeaker if it can be secured without at the same time producing undesirable cavity resonance effects. We know that the sounds coming from the rear of the cone are out of phase with those coming from the front; if, however, we can reverse the phase of the bass notes coming from the rear and allow them to emerge at the front of the loud-





speaker, we can secure desirable reinforcement of bass notes without increasing cavity resonance problems. The loudspeaker arrangement which accomplishes this result is known as a low-pass, phase-reversing acoustical filter.

The back of the cabinet or box is ordinarily closed completely when bass reinforcement is desired, and outlets for air are provided either underneath the cabinet or at the front, below the main outlet at the front of the cone. Our equivalent mechanical circuit for this condition is therefore the same as that shown in Fig. 10B. Again we can assume that undesirable effects of cavity resonance are greatly reduced by making the resonant frequencies of the cavity and the cone identical.

Above the resonant frequency of the cone or the cavity, the series resonant circuit $L_{\rm O}$ - $C_{\rm O}$ acts as a mechanical inductance, while the parallelresonant circuit $L_{\rm B}$ - $C_{\rm B}$ acts as a mechanical capacitance. To simplify further our analysis, let us assume an above-resonance condition (in which the frequency of mechanical force Fis higher than the natural resonant frequencies of the cone and box). In



FIG. 12. Equivalent acoustical circuit of a closed box baffle which is acting as a low-pass, phasereversing acoustical filter at frequencies above the resonant frequency of the cone, and vector diagrams showing how it acts.

this case, series resonant circuit $L_{\rm O}$ - $C_{\rm O}$, representing the cone assembly, will act as a mechanical inductance (which we can designate as L) in series with mechanical resistance $R_{\rm O}$; the equivalent mechanical circuit will now appear as shown in Fig. 12A.

If we can prove that mechanical current i_8 flowing through mechanical resistance R_B is 180° out of phase with mechanical current i_1 flowing through R_c , we will have shown that radiation of sound from the cavity will be 180° out of phase with the sound radiated by the back of the cone, and will therefore be in phase with sound radiated from the front of the cone. Above resonance, parallel resonant circuit $L_{\rm B}$ - $C_{\rm B}$ will act as a mechanical capacitance, and if by design its mechanical reactance is less than the mechanical reactance of L, the mechanical force F will feel an inductive load. The mechanical current i_1 delivered by F will therefore lag F by 90°.

A vector diagram will show at a glance the exact phase relationships between the various mechanical currents and voltages in our circuit. Let us use F as our reference vector, drawing it as shown in Fig. 12B. Since we have just found that i_1 lags F by 90°, we draw i_1 downward, 90° clockwise from F.

The mechanical force acting upon $C_{\rm B}$ and $L_{\rm B}$ will be equal to the applied force F minus the mechanical force dropped across inductance L. (We are now neglecting R_0 and $R_{\rm B}$, as they have little effect.) This mechanical force (mechanical voltage) $V_{\rm L}$ which is dropped across L will lead the current i_1 through L by 90° (coil voltage always leads coil current by 90°), so we draw vector $V_{\rm L}$ 90° counter-clockwise from vector i_1 , as in Fig. 12C. Now we can see that $V_{\rm L}$ is in phase with F.

We also know that the voltage across a condenser lags its current by 90°. Since parallel resonant circuit $L_{\rm B}$ - $C_{\rm B}$ acts as a condenser at frequencies above resonance, mechanical voltage $V_{\rm B}$ across this circuit will lag i_1 by 90°; we therefore draw in vector $V_{\rm B}$ 90° clockwise from i_1 , as in Fig. 12C. Since $V_{\rm B}$ acts upon $L_{\rm B}$, we can say immediately that mechanical current i_3 through L_B will lag V_B by 90°, and can draw in vector is 90° clockwise from $V_{\rm B}$, as in Fig. 12D. Our vector diagram now shows clearly that i_3 is 180° out of phase with i_1 , which is exactly what we desired to prove.

By proper selection of the mass (L_B) and compliance (C_B) of the box cavity in relation to the mass (L_C) and compliance (C_O) of the loud-speaker cone, an engineer can secure a loudspeaker system which will behave like the circuit in Fig. 12A. Sound waves escaping from the box will then be in phase with the sound radiated from the cone front, and the desired bass reinforcement will be secured at frequencies above the cavity resonant frequency.

At high sound frequencies, mechanical capacitance $C_{\rm B}$ serves as a shunt path for mechanical current, with the result that practically no mechanical current flows through $R_{\rm B}$, and no high-frequency sounds emerge from the box cavity.

Stromberg-Carlson Acoustical Lab*urinth.* An excellent example of the baffle design features just discussed is the acoustical labyrinth or winding passageway which is built into the cabinets of some Stromberg-Carlson receivers for four distinct purposes: 1, to prevent cavity resonance: 2, to prevent standing high-frequency sound waves; 5, to give a low cut-off frequency for the baffle without resorting to an excessively large cabinet; 4, to give reinforcement of bass response by allowing low frequency sounds radiated from the rear of the cone to travel through the labyrinth and emerge at the front in phase with the sounds normally radiated from the front of the cone. The two views in Fig. 13 illustrate the nature of this special loudspeaker baffle construction.

Observe that a felt hood covers the entire rear face of the cone, with only the pot exposed to air for cooling purposes. Sounds radiated by the rear of the cone are therefore directed into the rectangular cross-section passageway which winds back and forth down to the bottom of the cabinet. The insides of this passageway or labyrinth are lined with a porous soundabsorbing material which is held together by coarse metal screening. High-frequency sounds are totally absorbed by this material, while lower frequency sounds are only partially absorbed. The long passageway with its outlet at the far end has the essential features of mass, compliance and resistance which are necessary for a low-pass phase-reversing acoustical filter at bass frequencies, and the result is that bass sounds emitted at



FIG. 13. Essential constructional features of the Stromberg-Carlson Acoustical Labyrinth.

the lower end of the labyrinth reinforce the bass sounds radiated from the front of the cone.

Special RCA Box Baffles. To secure the same effects as are provided by the acoustical labyrinth, RCA engineers use the construction shown in Fig. 14A for some of their receivers. The back chamber is completely closed, but there are holes at the bottom of the chamber, each surrounded by a length of pipe, through which air can escape. The back chamber thus has mass and compliance, and that air which escapes through the pipe provides mechanical resistance. The pipes are intended to adjust the mass in order to make the chamber resonate at the same frequency as the cone. The entire cabinet of this RCA receiver is made of heavy wood, to prevent it from vibrating and radiating sounds from all sides when powerful bass notes are being reproduced. Only sounds emerging from the pipes are capable of reinforcing the normal output.

Pipes are by no means essential in the design of an acoustical low-pass filter; simple outlets are sufficient if the cabinet is properly designed to secure the correct resonant frequency for the cavity. Many RCA receivers have this simplified construction, illustrated in Fig. 14B. The back of the cabinet is closed by a curved sound reflector which serves two purposes, that of reflecting high-frequency sounds in all directions to prevent standing sound waves, and that of stiffening the back so it will not vibrate as readily as would a flat board. The air enclosed by this chamber provides compliance and mass, and the air escaping through the outlet at the bottom provides mechanical resistance; we thus have the conditions necessary for bass reinforcement.

Another example of this simplified construction is illustrated in Fig. 15. This is a special high-fidelity, highpower loudspeaker unit made by Jensen and known as the Peridynamic or bass reflex loudspeaker. The cabinet is made of thick, solid material so its sides will not vibrate. The box is made shallow so standing waves will be negligible. The cavity is properly proportioned to give the essential requirements of an acoustical filter.

Some manufacturers of radio receivers and cabinets follow the practice of closing up the back of the cabinet with a solid board which is reinforced with crossed ribs to prevent it from vibrating, allowing sounds to escape only from the corners of the back through holes which are left at these points. Again, proper design gives desirable bass reinforcement and eliminates undesirable features of a



FIG. 14. Constructional features of two types of low-pass, phase-reversing acoustical filters used in RCA receivers.

cavity. Do not think that because a loudspeaker cabinet appears simple in construction, it is easy to design and build; careful engineering is required in order to prevent cavity resonance, to prevent standing sound waves at high frequencies, and to secure the proper proportioning of acoustical elements in the cone and cavity in order to obtain the desired bass reinforcing action.



FIG. 15. Jensen Peridynamic loudspeaker unit. Reinforcing bass notes emerge from the rectangular opening below the cone.

Acoustic Clarifiers. When no attempt is made to utilize bass sounds emerging from the rear of a loudspeaker system, the chief problem is

that of removing cavity resonance effects or making them inaudible. It is possible to do this by inserting in the cavity an acoustic clarifier, a device which itself vibrates readily at all times and serves to cancel the air vibrations in the cavity. Some Philcoreceivers utilize this principle, in the form of a small cone which is suspended on an elastic spider mounted in the cabinet cavity or chamber. Any sudden loud sounds or sustained sounds which would otherwise set the cavity into vibration are transferred to this suspended cone, causing it to vibrate and absorb the energy present in the chamber. As a result there is very little vibration of the air in the chamber, and no sounds are emitted from the rear. Several of these acoustic clarifiers are generally used in a single cabinet, so as to increase the coupling between the cavity and the clarifiers. The clarifiers are sufficiently broad in frequency response to prevent cavity resonance effects when the back of the cabinet is open and the cabinet is placed flat against a wall. Remember, however, that clarifiers are not loudacoustic speakers.

The principle of an acoustic clarifier can best be explained by referring to the electrical circuit shown in Fig. 16. Here we have a parallel resonant circuit made up of L_1 and C_1 , driven by an A. C. voltage source V; this, as you know, is equivalent to the parallel resonant circuit representing the back chamber of a loudspeaker cabinet. At resonance, large currents flow through L_1 and C_1 , making this circuit act like a high resistance. If we place across this parallel resonant circuit a series resonant circuit made up of C_2 and L_2 , and make both resonant circuits tune to the same frequency, the series resonant circuit will at resonance act as a short across the

parallel resonant circuit. As a result, the source V will be supplying current to the series resonant circuit, while the parallel resonant circuit will receive practically no current. This series resonant circuit is the equivalent of the acoustic clarifier; it cancels the sound waves which otherwise would result in oscillation or vibration of the air in this back chamber.

Sound Diffusers

When sound is produced by a small source, all audio frequencies travel equally well in all directions away



from the source. When the size of the sound-producing source is quite large, however, as is the case with loudspeakers using large diaphragms and cones, only the low audio frequencies will travel equally well in all directions. The higher frequencies tend to travel best straight ahead of the loudspeaker, concentrating into a beam which becomes smaller and smaller in size as frequency is increased.

Loudspeaker engineers utilize what is known as a polar radiation pattern to show how a particular loudspeaker radiates various audio frequencies in different directions; one such pattern, representing the manner in which a cone loudspeaker with a baffle radiates one particular sound frequency in various directions, is shown in Fig. 17A. Let us first see how a pattern such as this is secured, for then we will be better able to appreciate the interesting information which it can give about a loudspeaker.

With our loudspeaker and baffle set up in a large room having soundabsorbing walls, floor and ceiling, and a definite audio frequency fed through the voice coil of the loudspeaker, we measure the loudness of the sounds produced by the loudspeaker at various points such as at P_1 , P_2 , P_3 and P_{4} , which are all the same distance away from the center of the loudspeaker and make various angles. such as θ (Greek letter theta), with the center line $O-P_3$. We will designate as Ia the loudness level measured at point P_8 . We now draw on naper a simple sketch of our loudspeaker, much like that in Fig. 17A, and mark point O as the approximate center of the sound-producing source. Point P_3 is placed on the diagram next, directly in front of the loudspeaker, and at any convenient distance away. A line is drawn from O to P_3 , and along this line we plot the measured value I_3 to any convenient scale, starting from O. We next move our measuring apparatus to position P_2 , which is off to one side of line $O-P_3$ by the angle θ but is still the same distance away from O. We measure the angle θ made by lines drawn from O to P_2 and P_3 , and use this angle as our guide in locating P_2 on the diagram. The loudness level measured at P_2 is now plotted along line $O-P_2$, starting from O; this gives point I_2 in Fig. 17A. The same procedure is repeated for points P_1 , P_4 and various other points which are all the same distance away from O and at various angles off to each side. A smooth curved line drawn through the resulting points I_1 , I_2 , I_3 and I_4 now gives the polar radiation pattern. In the case of Fig. 17A, this pattern tells us that at a definite distance away from the loudspeaker, sounds will be loudest directly in front of the cone and will gradually become weaker as we go off to one side or the other of the center line.

Radiation patterns for three different frequencies. as secured with a large dynamic loudspeaker mounted in a box baffle, are shown in Fig. 17B. All three curves are for measurements made at the same distance away from the loudspeaker. These curves tell that a listener at position P_1 would hear middle frequencies best, with bass frequencies slightly less loud and treble frequencies about half as loud. On the other hand, a listener at position P₂ would hear bass frequencies the best, with middle frequencies slightly weaker and treble sounds



FIG. 17. Polar sound radiation patterns for cone loudspeakers.

scarcely distinguishable at all. A line drawn from any other listening position to point O will give this same information for that position; simply observe where the line intersects each of the three curves. Now you can readily understand why the output from a loudspeaker sounds best when you are located directly in front of it.

Uniform distribution of sound from a loudspeaker throughout a room is highly desirable, for the non-uniform distribution shown in Fig. 17B tends to cancel out, for certain listening positions, all the advantages gained by careful design of the radio receiver and loudspeaker. Since high frequencies give the greatest trouble in this respect, the loudspeaker engineer often builds vanes in front of the loudspeaker to deflect these frequencies and spread out the beams. If the vanes are of the proper size, at least as wide as the wavelength of the high-frequency sounds to be deflected, the vanes will have no effectupon low and medium frequencies.

A wall-type loudspeaker, with deflecting vanes mounted in front of the cone for this purpose, is shown in Fig. 18A. Sometimes these vanes are mounted behind the grille cloth or are made a part of the cabinet itself.

A well designed deflector for highfrequency sounds is shown in Fig. 18B; this is a curved cone anchored rigidly inside the regular soundproducing cone of the loudspeaker. High-frequency sounds are deflected by this cone and spread out on all sides, while low- and medium-frequency sounds travel around the cone just as if it were not present.

Horn loudspeakers concentrate all sounds into relatively narrow beams, with the highest frequencies being concentrated the most. When a horn loudspeaker is used as a tweeter for the reproduction of high frequencies only, it is often sectionalized in the manner shown in Fig. 19 in order to spread out the beam and make the distribution of high-frequency sounds similar to that of the low- and medium-frequency sounds as produced by the woofer (low- and medium-frequency) loudspeaker. The presence of vanes in one or more units of a reproducing system is usually a sign of high-fidelity reproduction over a wide range of frequencies; this particularly holds true for the loudspeakers used in public address systems.

Horn-Shaped Baffles

Large cone loudspeakers are widely used in public address systems requiring high sound outputs. Cone loudspeakers mounted in a large flat baffle or in a box baffle radiate sound over a wide angle; this may not be



FIG. 18. Examples of cone loudspeakers which utilize deflecting vanes to spread out high-frequency sound waves.

desirable in some installations, for it is often necessary to concentrate sound in a definite direction and over a definite area. Horn-shaped baffles are used for this purpose; when properly designed, these make the loudspeaker more directional and at the same time improve its efficiency considerably.

Loudspeaker Efficiencies. The ordinary large cone loudspeaker, when mounted in a box baffle or on a flat baffle, rarely has an efficiency of greater than 3 per cent, whereas a driving unit used with an exponential horn can have an efficiency as high as 50 per cent. This means that if 10 watts of electrical power are fed into



FIG. 19. Horn loudspeakers designed for use as tweeters are often divided into sections as shown above in order to spread out high-frequency sounds.

a cone loudspeaker, only about .3 watt of acoustical power will be delivered. This low cone loudspeaker efficiency is satisfactory for the average home radio receiver, but in loudspeakers designed for high-power public address systems it is necessary to improve this efficiency by improving the coupling between the cone and the surrounding air. Loudspeaker engineers therefore turn naturally to the exponential horn when the efficiency of an ordinary baffle for a cone loudspeaker proves unsatisfactory.

An early attempt to improve the directional characteristics of a cone loudspeaker is shown in Fig. 20A; the performance of this flat-sided horn was no better than that of the early megaphone used with smaller driving units. Modern exponential horn

sounds back and forth inside the horn. The area of the mouth of the horn approaches the area of a circle whose diameter equals 1/4 wavelength at the lowest frequency to be reproduced.

Loudspeaker Impedance

Any loudspeaker has a definite electrical impedance. The value of this impedance will depend upon the frequency of the source voltage applied to the loudspeaker, upon the electrical characteristics of the loudspeaker (inductance and resistance for magnetic and dynamic driving units, and capacitance with resistance



FIG. 20. Examples of horn baffles for cone loudspeakers. A—simple flat-sided, megaphone-like horn; B—complete cone loudspeaker unit with true exponential horn baffle having a square cross-section; C another complete cone loudspeaker unit with an exponential horn of the trumpet type, spun from aluminum.

baffles or cone loudspeakers are shown in Figs. 20B and 20C. Since these start with a large throat, they need not be extremely long in order to secure good response over a wide range of frequencies. If the loudspeaker and cone are designed with a closed back, to give high air velocity at the front of the cone, and the usual precautions are taken to secure piston action over the entire audio range of frequencies, the addition of an exponential horn of this type will greatly improve the efficiency of operation.

In Fig. 21 is shown a large exponential horn which is fed by six large cone loudspeakers. Careful design is essential to prevent reflection of for crystal and condenser driving units), and upon the load which is placed upon the loudspeaker by the surrounding air.

We have already seen that the driving unit of a loudspeaker, when



FIG. 21. Two views of an unusually large exponential horn baffle which is fed by six large cone loudspeakers. This assembly is capable of delivering high sound output power at high efficiency. It is made by Lansing Manufacturing Company, Los Angeles, Calif.

in operation, has mechanical inductance, mechanical capacitance and mechanical resistance. Let us consider a simple loudspeaker in which these three mechanical components act in series upon a dynamic cone loudspeaker having electrical inductance and resistance. The complete circuit for this loudspeaker is best represented as in Fig. 22. Here ep represents the A.C. source voltage applied to the loudspeaker (this will be the A.C. output voltage of the audio amplifier), $r_{\rm P}$ represents the resistance of the source (the A.C. plate resistance of the output tube), $R_{\rm v}$ represents the resistance of the voice coil, $L_{\mathbf{x}}$ represents the inductance of the voice coil, M represents the electromechanical coupling, $L_{\rm M}$ the mechanical inductance (the mass of the moving element), $C_{\rm M}$ the mechanical capacitance (the compliance or springiness of the spider), and $R_{\rm M}$ the mechanical resistance. Clearly these mechanical components exist only when the driving unit is in motion; when the voice coil is wedged in position so it cannot move, the source would feel only the electrical components, R_v and L_v . By measuring the electrical characteristics of the loudspeaker under this condition, we can determine what these electrical values are.

When the voice coil and cone of the loudspeaker are free to move, and the loudspeaker is accomplishing useful work in converting electrical power into acoustical power or sound, the source feels all of the mechanical components just as if they were in an electrical circuit like that shown in Fig. 22. For example, if the source frequency is such that the mechanical reactance of $L_{\rm M}$ equals the mechanical reactance of $C_{\rm M}$, mechanical resonance occurs and only mechanical resistance $R_{\rm M}$ is reflected through M into the primary circuit. When the source frequency is above the resonant frequency of the secondary, the mechanical reactance of $L_{\rm M}$ exceeds that of $C_{\rm M}$, with the result that a mechanical inductance and mechanical resistance are reflected into the primary as a mechanical capacitance and mechanical resistance.

These facts are mentioned merely to show that when the electrical impedance of a loudspeaker is measured across its terminals (between terminals 1 and 2 in Fig. 22), a different value of impedance will be secured at each frequency; furthermore, this impedance will not be made up of voice coil reactance and resistance alone. but will also include those effects of the mechanical components which are felt by the source. The circuit in Fig. 22 tells us that the power input to the loudspeaker and consequently the useful sound power output will be a maximum when the source impedance (resistance) $r_{\rm P}$ is exactly matched to the load impedance (the loudspeaker input impedance) as measured between terminals 1 and 2.

Clearly the matching of the loudspeaker impedance with that of its voltage source is quite important if greatest sound power output is to be obtained. The circuit in Fig. 22 shows that the D.C. resistance R_v of the loudspeaker, which can be measured with an ohmmeter, cannot be considered alone when securing a proper match; it is the loudspeaker impedance as measured while the loudspeaker is actually in operation which must be matched to the source impedance.

Determining Loudspeaker Impedance. Unfortunately the impedance of a loudspeaker is not constant for all frequencies, nor is the mechanical load of a loudspeaker as simple as is represented electrically in Fig. 22. For these reasons it is necessary to measure the loudspeaker impedance at the frequency which we wish to reproduce at greatest volume, and secure a correct impedance match for this frequency.

For the average loudspeaker an impedance match at about 1.000 cycles is usually quite satisfactory. With tweeter loudspeakers, however, match at 4,000 cycles would be better. Loudspeaker manufacturers recognize these problems, and specify a value of loudspeaker impedance which, if matched, will give the best over-all results over the frequency range to be handled. For example, when the impedance of a loudspeaker voice coil is specified as 8 ohms, this will take into account mechanical as well as electrical factors and will probably be an impedance measurement made while the loudspeaker was reproducing a 1,000 cycle audio signal.

When the impedance of a dynamic loudspeaker is unknown, some servicemen measure the voice coil resistance with an ohmmeter and assume an impedance value equal to 1.5 times this measured D.C. resistance. We know that this value can be considerably in error, but it at least serves as a guide when a more accurate impedance value is not available.

Matching. Knowing the impedance of the loudspeaker at the average or predominant frequency in the range of audio sounds being handled, the next step is to match this impedance to the impedance of the electrical signal source, in order to secure maximum useful sound output. When vacuum tube amplifiers, which are the usual signal sources for loudspeakers, feel anything different from the correct, properly matched load, the wave form of the signal fed to the loudspeaker will be distorted. The reason for this is simply that with an improper impedance match the output tube of the amplifier no longer operates over the linear portion of its dynamic E_{g} - I_{p} characteristic. Regardless of whether the output stage uses class A, pushpull or push-push operation, best results are obtained when the load placed on the stage has the correct impedance value.

With the loudspeaker impedance and the required load impedance of the amplifier known, there remains only the connecting together of source and load by means of a suitable matching or output transformer. Since the re-



FIG. 22. Equivalent circuit of a complete loudspeaker unit connected to a source of audio frequency power.

quired plate load impedance is usually much greater than the loudspeaker impedance, a step-down transformer is needed: its turns ratio is determined simply by dividing the required plate load impedance by the loudspeaker impedance and then taking the souare root of the resulting number. (Taking the square root of a number means finding a smaller number which, when multiplied by itself, will give the original number.) The primary winding, which connects into the plate circuit of the output tube or to the plates of the tubes in a push-pull or push-push output stage, will have the greatest number of turns and will therefore usually have the greatest D.C. resistance. The secondary winding has the least turns and connects to the loudspeaker voice coil leads. Here is an example: If a push-pull output stage requires a 2,000-ohm load for greatest undistorted output power, and the loudspeaker has a specified impedance of 20 ohms, our impedance ratio is 2,000 divided by 20, which is 100. Taking the square root of 100, we get 10 as the turns ratio of the output transformer. The primary winding Typical Loudspeaker Matching Circuits. A typical connection of a loudspeaker and output transformer to a single tube output stage is shown in Fig. 23A. The recommended plate load impedance of the pentode output tube and the average voice coil impedance of the loudspeaker determine-



These diagrams illustrate a number of the interesting features of cone loudspeakers which are taken up in this lesson.

A—Balanced armature driving unit which is coupled mechanically to a medium-sized free-edge cone. This unit is commonly known as a magnetic loudspeaker.

B and C-Typical small cabinets used for magnetic loudspeakers.

D-Permanent-magnet dynamic cone loudspeaker; this is similar in appearance to many electrodynamic loudspeakers, but the trade name NOKOIL indicates a permanent magnet construction.

will therefore have 10 times as many turns as the secondary winding. With the loudspeaker connected to the secondary winding and an audio signal of average frequency being fed into the primary winding, the measured impedance of the primary will be 2,000 ohms, and we therefore have a correct impedance match. E—Special form of dynamic loudspeaker employing a para-curve or curvalinear cone in order to provide uniform response over a wide range of frequencies.

F-Dynamic loudspeaker employing a corrugated cone construction for uniform response over medium frequencies; the conical plug over the moving coil in the center of this cone serves to diffuse high frequency sounds in all directions and to increase the frequency range over which piston action is secured.

G-Special dynamic loudspeaker cabinet designed for wall mounting. Magnetic loudspeakers are also available in cabinets like this.

the turns ratio required for the output transformer.

A typical loudspeaker connection for a two-tube output stage is shown in Fig. 23B. With class AB or A' push-pull operation, a C bias resistor must be inserted in the circuit at point X to provide a negative C bias for each tube, and this resistor must be shunted with a high-capacity condenser. Since only one tube works at any instant of time, and feeds into only one-half of the output transformer primary winding, we must multiply the recommended plate load impedance for one tube by 2 in order to determine the turns ratio for the entire output transformer.

With high-impedance loudspeakers, such as those employing magnetic, condenser or crystal driving units, the loudspeaker impedance may be very close to that required by the output tube. The loudspeaker may in this case be connected as in Fig. 23A, using placement transformer in every case in order to insure a correct match and duplicate the original performance of the receiver. This is particularly important in the case of receivers or public address amplifiers designed for high-fidelity operation.

When exact replacement transformers are not available, you can make a satisfactory replacement with little apparent change in performance by selecting a transformer having the correct turns ratio and a sufficiently large power-handling rating. Few radio men are able to order transformers on this basis, however; for



FIG. 23. Typical loudspeaker connections to output stages. Condenser Co is often included to prevent self-oscillation of the output stage, to alter the frequency response of the amplifier, or for both reasons.

a 1-to-1 turns ratio output transformer (having the same number of turns on the primary as on the secondary), or a direct loudspeaker connection to the plate circuit may be made in the manner shown in Fig. 23C. In this latter case the D.C. plate current of the output tube flows through iron-core choke coil CH to the power supply, while signal currents flow through D.C. blocking condenser $C_{\rm R}$ and the loudspeaker. The inductance of choke CH must be sufficiently high to prevent excessive bypassing of audio frequency current around the loudspeaker.

Replacement of Output Transformers. When output transformers become defective and must be replaced, it is best to use an exact rethis reason radio supply houses list replacement transformers for use with particular output tubes and for either single- or two-tube output stages. These transformers are designed on the assumption that most dynamic loudspeakers have a normal voice coil impedance of about 10 ohms.

Radiotricians oftentimes prefer to use universal output transformers, a few of which they can keep on hand at all times. These transformers are available in various power ratings, and have a number of taps on the secondary to provide various turns ratios. The primary winding has a center tap which is used with twotube output stages but is ignored in the case of a single-tube output stage. The taps on the secondary usually provide for voice coil impedances of from 1 to 30 ohms. Use that tap which gives the greatest volume along with clearest tone when an actual radio program is being reproduced. For a more accurate determination of the correct tap, connect a cathode ray oscilloscope to the voice coil terminals, feed into the audio amplifier a pure sine wave signal first at 100 cycles, then at 1,000 cycles and 4,000 cycles, and select that tap which gives the greatest output voltage along with sine wave output for all three of these frequencies.

Field Coil Connections for Electrodynamic Loudspeakers

Most of the dynamic or moving coil loudspeakers used in radio receivers are of the electrodynamic type, employing a field coil for the production of a fixed magnetic flux. The average power used is from 7 to 15 watts. In many cases these coils are made to serve also as chokes for the power pack system, for at power line frequency the field coil has an appreciable inductive reactance. When used in this manner in place of a filter choke, the D.C. resistance of the field coil becomes important, for this resistance reduces the net D.C. supply voltage which is available for the tubes in the receiver.

The loudspeaker field coil connection shown in Fig. 24A is without doubt the most widely used connection today in radio receivers. The D.C. resistance of a field coil used in this manner will usually be somewhere between 500 and 2,000 ohms. Since the D.C. current drawn from the power pack by the various receiver circuits must pass through this field coil, the coil must be capable of handling this current without excessive overheating. When the receiver has a push-pull output stage, complete filtering of the plate and screen grid voltages for the stage is not necessary, and a somewhat higher D.C. voltage is obtained by connecting to point X rather than to B+.

Since the power pack output voltage in a universal A.C.-D.C. receiver is considerably lower than that in an-A.C. receiver, any loudspeaker field coil connection which would reduce this output voltage even more is obviously not desirable. For this reason the field coil of a loudspeaker in a universal receiver is usually independently supplied with rectified current in a manner similar to that shown in Fig. 24B. In this circuit, one half of the 25Z5 double diode rectifier tube supplies current for the field coil, with filter condenser C_1 removing the pulsations in the half-wave rectified output current. The other diode section of this tube likewise delivers half-wave rectified current which is smoothed by filter condenser C_2 and choke coil CH. This choke coil is sometimes replaced by a resistor or is omitted entirely, and a higher-capacity filter condenser is used to provide the necessary filtering. The D.C. resistance of the loudspeaker field coil is usually larger than 2,000 ohms in a circuit of this type, in order to limit field coil current and prevent overloading of the rectifier.

In addition to providing a fixed magnetic flux and filtering the power pack output current, a loudspeaker field coil is sometimes made to serve a third purpose, that of providing a negative C bias voltage for the receiver stages. A tap is provided on the field coil for this purpose, as indicated in Fig. 24C; the value of the bias voltage produced depends upon the resistance which is present between the B— terminal of the power pack and the tap, and upon the amount of D.C. current flowing through this resistance. The entire D.C. voltage drop across the loud-speaker field coil can also be utilized for C bias purposes; the C--terminal in Fig. 24C thus provides a greater negative bias voltage than the C--terminal.

Separate Field Supplies. Loudspeakers used with public address amplifier systems must often be located a considerable distance away from the Fig. 24E. The vacuum tube rectifier normally supplies an output voltage of over 250 volts, and the field coil used with this supply must therefore have a high resistance in order to limit the current to a safe value. With the dry disc type of rectifier, the output voltage will usually be less than 100 volts, and a lower-resistance field coil can be used. The present trend is toward use of permanent magnet



FIG. 24. Typical electrodynamic loudspeaker field supply connections.

amplifiers. If field coil leads were run from the loudspeaker location to the amplifier, the resistance in these leads would be excessive. The same holds true for extra loudspeakers, which are often connected to radio receivers but located some distance away from the receiver. In cases like these, it is necessary to use a separate field supply located at the loudspeaker if it is of the electrodynamic type. This field supply may consist of a full-wave vacuum tube rectifier like that shown in Fig. 24D, or may be a full-wave bridge type rectifier using copper-oxide discs, as indicated in

dynamic loudspeakers, however, thus eliminating entirely the need for a separate power supply at the loudspeaker location.

With battery receivers the extra current drain due to the field coil of an electrodynamic loudspeaker would obviously be undesirable; for this reason the loudspeakers used in battery receivers will generally have either a balanced armature type magnetic driving unit or a permanent magnet dynamic driving unit. Auto radio receivers use electrodynamic loudspeakers extensively, with the field coil so designed that it can be connected directly to the 6-volt storage battery; permanent magnet dynamic loudspeakers are also used with auto radios.

Þ

Elimination of Hum. We have seen how the rectified current supplied to the field coil of the loudspeaker in an A.C. receiver is usually incompletely filtered. Up to a few years ago this ripple current was not objectionable because the average loudspeaker was not able to reproduce sounds as low as the hum frequency of 120 cycles. Recent improvements in loudspeaker design have extended the low frequency response of the loudspeaker well down to the lowest audible sound frequencies; the result is that any ripple currents induced in the voice coil are heard as objectionable hum. This condition arises because magnetic flux produced by the field coil passes through or links with the voice coil, giving the effect of a transformer in which A.C. or ripple current going through the primary induces corresponding A.C. voltages in the secondarv or voice coil. The two basic methods used for eliminating hum due to A.C. in the field of a dynamic loudspeaker are: 1, use of a shading ring; 2. use of a hum-bucking coil.

Shading Ring. A large copper ring called a shading ring or shading coil is placed around the central core, as shown in Fig. 25A. This shading coil acts as a short-circuited turn which, when A.C. currents flow through the field coil, sets up an out-of-phase A.C. flux in the core to cancel out the effects of the original A.C. flux. The result is that no A.C. voltages are induced in the voice coil.

Hum-Bucking Coil. A more widely used scheme, shown in Fig. 25B, involves winding on the central core a number of turns of wire (usually a few less than are in the voice coil). This extra coil, known as a humbucking coil, is connected in series with the voice coil but is wound in the opposite direction. Any A.C. flux which is present in the core induces an A.C. voltage in the voice coil and also in the extra coil. Since these voltages are out of phase and acting in series, they balance or buck each other, and no A.C. current flows through the voice coil to react with the permanent magnetic flux. Circuit diagrams usually indicate when a



FIG. 25. Here are two methods used hy loudspeaker design engineers for overcoming hum which is due to the presence of A.C. components in the field coil current of an electrodynamic loudspeaker.

hum-bucking coil is present in the loudspeaker; typical schematic circuit diagrams illustrating this hum-elimination feature are shown in Fig. 26.

Acoustical Problems of High-Fidelity Reproduction

The output of the average loudspeaker begins to drop rapidly at frequencies above 4,000 cycles. This drop in high-frequency output can be offset to a certain extent by designing the audio amplifier to have a peak response in the high-frequency region, but there is a limit to the amount of treble-boosting which can be incorporated in the audio system without introducing new circuit problems. This is one of the reasons why modern high-fidelity loudspeaker systems usually employ two separate loudspeakers, a woofer, which is capable of giving flat response up to about 3,000 cycles, and a tweeter, which provides satisfactory reproduction of the higher frequencies. Typical response curves for a woofer and a tweeter are shown in Fig. 27. The



FIG. 26. Schematic circuit diagrams of radio receivers usually indicate, in a manner similar to that shown above, whether a hum-neutralizing (humbucking) coil is used in the loudspeaker.

dotted line gives their combined response over the region in which they overlap, and shows that the combined effect is a greatly increased range over which flat response is secured. A lowpass filter (often simply a shunting condenser) is placed in the woofer circuit and a high-pass filter is placed in the tweeter circuit (a series condenser will often suffice) to prevent useless currents from overloading the loudspeakers.

In public address systems, both woofer and tweeter may be of the horn type, or the woofer may be a dynamic cone loudspeaker using an exponential horn baffle, with a horn type tweeter. Deflecting vanes are often built into the tweeter to broaden its directional characteristics.

In home radio receivers the woofer



is usually a dynamic cone loudspeaker mounted in a box baffle having acoustic low-pass phase-reversing construction, with a tweeter loudspeaker of either the cone type using a crystal driving unit or the trumpet horn type also mounted on the baffle. A typical woofer-tweeter combination in a special box baffle is shown in Fig. 28.

Room Acoustics. Although most people recognize the fact that a theatre or auditorium must be properly treated with sound-absorbing material in order to give natural reproduction of sounds, the average radio receiver owner gives very little attention to the acoustic treatment of the room in which his radio receiver



FIG. 28. A large dynamic cone loudspeaker (woofer) and a horn type tweeter are here mounted on a single box baffle to give uniform response over a wide range of audio frequencies. The inside of the box is lined with sound-absorbing material to prevent standing waves.

is located. It is the duty of the Radiotrician to point out that a room having overstuffed furniture, heavy draperies, and a large, heavy rug is desirable. There will then be sufficient sound absorption to prevent sounds from being reflected back and forth between the walls of the room and various objects in it, and there will be no echo or reverberation The Radiotrician can also effects. recommend that the loudspeaker be pointed or directed in such a way that sound can travel a maximum distance before being reflected. If the radio is in one corner of a room, the receiver should be facing another corner

World Radio History

diagonally opposite, giving a long path through air for the radiated sounds.

High-frequency sounds are so directional in nature and so easily absorbed by rugs and furniture that they should be heard directly rather than after reflection. Some receivers utilize inclined sounding boards or baffles which cause these high-frequency sounds to be directed at an upward angle, away from the floor, for this purpose.

By this time it should be quite clear to you that the securing of highfidelity reproduction in a radio receiver system or public address system can be best obtained only by proper design of each and every link in the long chain between the microphone in the broadcast studio and the listening location in the home.

Loudspeaker Replacement Hints. Tone controls can compensate to a

certain extent for deficiencies in loudspeaker response, but manufacturers of high-quality receivers will sometimes introduce equalizer circuits in an audio amplifier in order to make the amplifier response peaked in a region where the loudspeaker_response is low. The combination response of the system can in this way be made almost uniform for high sound levels. Tone controls then correct for the peculiarities of the human ear at low sound levels. Changing of loudspeakers may therefore alter the performance of a highfidelity receiver, unless an exact duplicate replacement loudspeaker is used. When the loudspeaker cone in a high-fidelity receiver must be replaced, always use the exact duplicate replacement cone which is supplied by the receiver manufacturer; cone design has considerable influence on loudspeaker response.

Lesson Questions

Be sure to number your Answer Sheet 27FR-2.

Place your Student Number on every Answer Sheet.

Send in your set of answers for this lesson immediately after you finish them, as instructed in the Study Schedule. This will give you the greatest possible benefit from our speedy personal grading service.

- 1. Why are large flat diaphragms unsatisfactory for loudspeakers? They will bend or Buckle ,
- 2. What two purposes does a spider serve in a dynamic loudspeaker? Center the voice Coll, Retwining the brice toil 3. Does mechanical resonance occur at high, medium or low frequencies in
- dynamic cone loudspeakers? Low prequencies
- 4. Why is pure piston action desirable in the cone of a dynamic loudspeaker? To secure a flather prequiricy response
- 5. Is an irregular shape of baffle better than a square or found baffle? yes,
- 6. Can the natural resonant frequency of a box baffle be lowered by increasing the size of the box? Yes
- 7. How can the undesirable effects of cavity resonance be reduced in a box type baffle? Larger bay or infinite flat baffle
- 8. What are the four distinct purposes of the acoustical labyrinth? 9. What are the approximate maximum obtainable efficiencies for a large
- cone loudspeaker in a box baffle and for a driving unit used with an ex- 50ponential horn? 370
- 10. What are the two basic methods of eliminating hum due to A.C. in the field of an electrodynamic loudspeaker? Shading Ring 29 them buckling Coil

DELIVER THE GOODS

"There are 57 rules for success. The first is to deliver the goods. Never mind the rest."

Like many striking assertions, the quotation above is not altogether true, because there are other rules which cannot be ignored. But there is a lot of truth in this statement.

If you want to be a success in life, deliver the goods. Employers want men who *earn* their salary each and every day—men who have the training required for their particular jobs and actually do apply this training to their work.

You can always excuse yourself if you fail—but nobody else will ever excuse you. Customers may be polite to you and may feel sorry for you, but they will go elsewhere the next time they need radio service work. Employers are equally indifferent to excuses, for they must have good men if they themselves are to deliver the goods.

Make it your business to be where you are needed, when you are needed, with the service or help that is needed. Have all the knowledge which may be required. Be the man who delivers the goods and gets the money.

J. E. SMITH



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28FR-3

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STUDY SCHEDULE NO. 28

For each study step, read the assigned pages first at your usual speed. Reread slowly one or more times. Finish with one quick reading to fix the important facts firmly in your mind, then answer the Lesson Questions for that step. Study each other step in this same way.

The most widely used measuring device is the D'Arsonval d.c. meter. Learn its principles and all other meters are easy to understand. Here you learn how the meter range is extended and how meters are calibrated. This section is full of practical information on using current meters-information which may prevent your damaging your equipment. Answer Lesson Questions 1, 2, 3 and 4.

There are many ways of measuring alternating current. These methods have advantages and disadvantages which you must keep in mind; otherwise, you can get misleading readings. Here the principles of these methods are fully developed. Pay particular attention to the information on the copper-oxide rectifier, as this is the most widely used a.c. rectifier. Answer Lesson Question 5

Voltages are measured by the same kind of meter used for current. The important difference is the arrangement of resistors used with the meter. A voltmeter will not always indicate the true operating voltage, depending on its sensitivity. The reason for this, as well as the practical means of allowing for this condition, is clearly brought out by practical examples. You won't be confused by your voltmeter readings after reading this section. Answer Lesson Ouestions 6 and 7.

The ohmmeter is probably the most used service instrument. With it you determine resistor values, check condensers for shorts or leakage and check coils and circuits for continuity. Practical data on using and safeguarding this valuable device is given. Answer Lesson Questions 8 and 9.

The multimeter is a combination current meter, voltmeter and ohmmeter, built into a single unit for convenience in using. The two basic types are covered, so you will know how to use either. Of course, there are many different scale ranges available, as each manufacturer has his own ideas as to the most desirable ones. Practically all the standard makes are good, however, once you learn to read their scales. This section closes with an important review of meter tolerances and their practical importance. Answer Question 10.

🛐 6. Mail your Answers for this Lesson to N.R.I. for Grading.

🚺 7. Start Studying the Next Lesson.

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CURRENT, VOLTAGE AND RESISTANCE MEASUREMENTS

Direct Current Meters

METERS are used constantly in radio work of all kinds. As a radio serviceman, you'll make at least a few meter measurements on practically every set you handle.

Early in your N.R.I. Course you learned the elementary facts about what these instruments are and how they are used to measure current, voltage and resistance. In this lesson you'll learn the practical details of actual commercial meters—how they're made, how they work, what to expect from them, and—most important of all the correct ways to use them.

HOW IS CURRENT MEASURED?

We cannot see an electric current, but we can detect a number of effects produced by its flow. We might use any one of these effects to measure the amount of current flowing.

For example, we might use a thermometer to measure the heat produced by a current flowing through a wire. From this heat measurement, we could calculate the amount of current.

Or, we might send the current through a resistance wire that will become hot enough to give off light—a common light bulb, for example. By measuring the amount of light produced, we could determine the amount of current flow.

Again, we might make the current flow through a certain silver salt solution, causing a silver electroplating action. By weighing the amount of silver thus "plated," we would have another means of measuring current. (This last, under carefully controlled conditions, is an international standard method of determining exact current flow.)

But these methods are generally too slow and cumbersome for practical uses. We need a device which will react at once to changes in current and which can be used anywhere, not just under laboratory conditions.

One of the first electrical discoveries was that the magnetic field which always accompanies a current flow would make a compass needle move. Experiments soon proved that the distance the needle moves depends on the amount of current flowing. This was a highly important discovery, for it pointed out the basic principle used in most modern meters. Let's now see how these meters use this magnetic effect to measure current.

THE D'ARSONVAL METER

Early experimenters soon found that the compass needle method of measuring current was not accurate enough, since it depended on the amount of magnetism in the needle and the effect of the earth's magnetic field. A meter (known as the D'Arsonval type, after its inventor) was invented which overcame these objections.

Instead of placing a compass within
a coil, a small coil is suspended so it can move, and is placed in a strong magnetic field. When the current to be measured flows through the coil, a magnetic field is set up which interacts with the fixed field, causing the coil to move. The basic principle of this operation is shown in Fig. 1. You remember that like magnetic poles will repel. If a small pivoted magnet is placed as shown in Fig. 1A, it will move



FIG. 1. The coil has a magnetic field and will move when placed in a fixed field.

in the direction of the arrow. A current-carrying coil, such as the one shown in Fig. 1B, has a north and a south pole. If this coil were put in place of the small magnet in Fig. 1A; it too would move.

Unless some means are taken to stop it, once the coil starts to rotate, it will continue to do so. That is why the two fine spiral springs are attached at the ends of the coil. When the coil moves, the springs are twisted so that they oppose the coil movement. The more the coil rotates, the more opposition the springs give. Thus, the coil moves only to the point where the magnetic force causing rotation is balanced by the retarding action of the springs. When the current stops, the magnetic field of the coil disappears, and the springs move the coil back to the starting position.

The magnetic force causing rotation of the coil is proportional to the amount of current flowing through the coil. Therefore, one particular value of current will make the coil rotate to one particular place. A larger current will rotate the coil farther; a smaller one will rotate it less. By attaching a pointer to the coil, the amount of movement can be indicated on a scale. This scale, when properly marked, will show us the amount of current flowing through the meter.

This is the basic principle of the D'Arsonval type meter movement. It has been developed into the modern, practical meter shown in Fig. 2. Let's look at its features.

The small, powerful magnet is made from special steel or metal alloys, chosen for strong magnetic qualities and long magnetic life. The stronger the fixed field, the more the coil will move for a particular current, so the permanent magnetic strength is made as high as possible. The magnet is specially treated and aged until the field strength remains constant.

The pole pieces are soft iron, carefully shaped to give the desired magnetic field distribution. If the meter scale is to be linear (that is, so .adjusted that equal increases in current will produce equal increases in meter movement), the magnetic field must be uniform through the gap in which the coil turns. A soft iron core is inside the coil form. The coil moves through a small gap between this core and the pole pieces. By making this gap small to reduce the reluctance of the magnetic path, we obtain a strong field.

To make the coil easy to move, it is wound on a very light, thin metal form, and the coil and form are suspended between almost frictionless pivots with jewel bearings. The number of turns and size of wire used to make the coil depend on the current range and sensitivity desired for the meter.

The coil must start to move from

the same position each time, in order for the pointer to come to rest at the proper point on the meter scale. This starting, or zero, position is determined by the springs. They are wound in a loose spiral so that they oppose any rotation which would either wind them tighter or unwind them. When the coil moves, one is wound while the other is



FIG. 2. The names of the meter parts.

unwound. The springs thus oppose a coil movement in either direction away from the starting position, which is the position the pointer takes when no current is flowing in the coil.

Naturally, these springs won't always remain perfectly balanced. That is why practically all good meters have a zero adjustment. This is a small screw which usually protrudes through the case of the meter just below the meter coil. By turning this screw, the upper coil spring can be moved enough to balance the springs and bring the meter pointer back to the zero position.

The springs are also used to make the electrical connections to the coil, as shown in Fig. 1B. The springs are metal, so they make ideal leads. Of course, this means they must be insulated from each other and from the meter frame.

Meter Damping. When a current flows through it, the coil (because of. its inertia) tends to overshoot or move past the correct point a small distance. Then the springs pull the coil back on the other side of the correct point. In other words, there is a wavering back and forth for a time before the meter pointer settles down to the proper readings. To keep this wavering as small as possible, the meter coil frame is made of metal. When it moves through the permanent magnet field. eddy currents are induced in the frame. These eddy currents set up an opposing field which slows down the rate of coil movement, so that the coil comes to the correct point and stops. This steadying action is called damping.

Another common method of damping is to place a resistor between the meter terminals. A voltage is induced in the coil as it moves through the fixed field. This induced voltage causes a current flow through the resistor and coil, which sets up an opposing field with a result like that caused by the eddy currents induced in the coil form. In both cases, the damping action ceases as soon as the coil stops moving.



What the inside of a meter looks like.

The resistor value which will permit the most rapid coil movement, without noticeable wavering at full-scale meter reading, is called the *critical damping* value. This value varies widely with different meters—some may require 10,000 ohms, while others may need as low as 100 ohms. A resistor below the critical value causes overdamping and a slow meter movement, while too large a resistor does not damp enough. / meters.

The eddy current method of damping does not affect the meter range at all. The resistor method may or may not affect the current range of the meter, depending on the value of the resistance needed. You will learn more about this later on in this lesson.

CALIBRATION

The meter we have described will indicate when current is flowing. However, we are usually most interested in the actual amount of current flowing, so we want the meter to have a scale marked off in units of current flow.

By passing known currents through a meter, we can mark the scale according to the deflection obtained and thus calibrate the meter. "Calibrating" is the technical term for finding the accurate values of points on the scale of an instrument. For example, suppose you had a stick of wood with scratches on it that you wanted to use as a ruler. You'd "calibrate" it by finding how many inches each scratch was from one end of the stick. Meters are calibrated the same way-except, of course, each point on the scale is marked in terms of the current, voltage or other quantities needed to make the meter needle swing to that point.

Large laboratories, such as the Bureau of Standards, determine currents very accurately by chemical or other means and hand-calibrate a standard meter. Then the standard meter is placed in series with a meter to be calibrated, as shown in Fig. 3. The amount of current passing through the two meters is varied in steps, and the meter readings are noted for each step. Since the standard meter is indicating the exact current flowing through both meters, the other meter can be directly calibrated from the standard. This method is used in the manufacture of the more accurate and expensive meters.

Meters of reasonable accuracy and much lower in price can be made by duplicating the hand-calibrated meter as closely as possible in all parts. These production meters usually have about the same response as the meter on



FIG. 3. Calibrating a meter by comparison.

which they were modeled, so the same scale can be used for all of them with fair accuracy. To improve the accuracy of their mass-produced meters, manufacturers adjust the spring tension and magnetic field so that the fullscale reading of each meter is correct.

Percentage Accuracy. Highly accurate, hand-calibrated meters are expensive, and not necessary for service work or ordinary testing. Servicemen generally use production meters with printed scales. The percentage of accuracy of such meters is expressed in terms of full-scale deflection. For example, suppose a milliammeter has a full-scale reading of 100 ma. If it is off 5 ma. when the needle gives a fullscale deflection, the meter is said to have a 5% accuracy. (5/100 - .05 or 5%). In other words, when the needle points to 100 ma., the actual current being measured may be 95 ma. or 105 ma., or any value in between.

Now this 5% accurate meter may have much poorer accuracy for readings at the low end of the scale. When the meter points to 10 ma., the meter may still be wrong by 5 ma., which would mean a 50% error. (5/10 - 0.5)or 50%). Of course, most of the better meters are held within 1% or 2% of full scale, and the inaccuracy at the low end of the scale is not as great as that given in our example. But here's a practical tip—if several meter ranges are available, use the one giving a deflection of more than half-scale. Then your reading will have the loast error.

Parallax. Inaccurate readings may result from looking at a meter scale from an angle rather than from right above it. Figure 4 shows you why.

In order to swing freely, the pointer has to be at least a little distance above the scale. Now when you look right down on the pointer, you get one reading. Say it's 10. If you are off to one side, the pointer will seem to give a different reading. It might be 9 or 11, depending on which side you are. The farther you are to either side, the farther off the apparent reading will be from the true onc. (You can prove this for yourself by trying to read your watch at an angle.) This shift in the apparent position of the pointer, due b



FIG. 4. Always look straight at a meter to avoid parallax.

to the position of the observer, is called parallax.

Most laboratory meters have a small mirror on the scale under the pointer, so unless your eye is directly above the pointer, you will see a reflection of the pointer in the mirror. With such meters, you can avoid parallax by always moving to where the reflection disappears behind the pointer before making the reading.

In service and communications

work, highly accurate readings are usually unnecessary. Even so, good service meters have pointers shaped like knife blades. These pointers look thinnest when viewed from the correct position.

EXTENDING CURRENT RANGES

In practical service and laboratory work, both large and small currents have to be read. We could use a number of different meters having the re-



FIG. 5. The current flow divides between the meter and shunt.

quired ranges, but meters are costly and bulky. It is far better to have one meter and a means of changing its current range.

Shunts. Suppose we have a 1-ma. meter and want to measure 10 ma. We can do so by putting a resistor across the terminals of the meter of such a value that 9/10 of the current coming into the resistor-meter combination will flow through the resistor and only 1/10 will flow through the meter. In other words, we use the resistor to "bypass" 9/10 of the current. Figure 5 shows how this is done. When we read the meter, we simply remember that there is 10 times as much current flowing as the meter shows—so we multiply the meter reading by 10 to get the actual current flow. If our meter reads 0.5 ma., we know the real current is 5 ma. $(0.5 \times 10 - 5)$; if it reads 1 ma., the actual current is 10 ma. (1×10) = 10).

Since the "by-pass" resistor R makes

a parallel (or shunt) path around the meter, it is called a shunt. The ohmic value of R is calculated so that it will pass a current equal to the difference between the total current to be measured and the amount the meter needs for full-scale deflection. For example, suppose we want to measure currents up to 25 ma., but all we have is a 1-ma. meter. Then we'll have to find a shunt R which will pass 24 ma.

To do this, we use Ohm's Law, R - $E/I_{\rm s}$. The voltage E is the voltage across the meter terminals. This voltage is called the meter millivolt rating and is equal to the basic meter current range multiplied by the meter resistance, so we can substitute for E in our Ohm's Law equation the quantity $I_{\rm M} \times R_{\rm M}$. ($I_{\rm M}$ is the basic meter current; R_{M} is the meter resistance.) The current I_s is the current which passes through the shunt. This current equals the total current minus the basic meter current, so we can substitute for I_s the quantity $I_T - I_M$ (I_T is the total current flowing; $I_{\rm M}$, as before, is the basic meter current.)

Now our Ohm's Law equation reads:

$$R = \frac{R_{\rm M} \times I_{\rm M}}{I_{\rm T} - I_{\rm M}}$$

We can find R by substituting values for the other terms. (All currents must be figured in the same units—that is, all in milliamperes or all in amperes. Then R will come out in ohms :

Suppose our 1-ma. meter has a resistance of 100 ohms. The current $I_{\rm T}$ we want to measure is 25 ma. Then,

 $R = \frac{100 \times .001}{.025 - .001} = \frac{.1}{.024} = 4.2 \text{ ohms,}$

and we've found the size shunt we need.

To find the actual total current flowing, multiply the reading on a shunted meter by the ratio of the current range of the meter with shunt to the current range without shunt. In our example, a 1-ma. meter was made a 25-ma. meter by the use of a shunt, so meter readings must be multiplied by 25 to learn the actual current flow (25/1 - 25). Remember—the meter itself is not passing 25 ma. Only 1 ma. goes through the meter, and 24 ma. go through the shunt.

You remember that a shunt resistor helps damp a meter. Most shunts have a resistance less than the critical damping value, so the meter movement is slowed down somewhat. This overdamping is not usually objectionable, however.

MULTI-RANGE CURRENT METERS

Very likely the meter you'll use in your service work will have a number of ranges, with several built-in shunts and a switch arrangement. Figure 6 shows one way such a meter might be made. When the selector switch is in position 1, the meter is not shunted. In other switch positions, different shunts are thrown in, giving various ranges. Typical ranges might be: 1 ma.; 10 ma.; 100 ma.; 10 amperes. Whatever



FIG. 6. Multi-range current meter.

the ranges, the manufacturer will mark the scale for each range so you won't have to do any figuring. The only usual exception is when one range is 10 times another, as when the ranges are 10 ma. and 100 ma. There, you may have to multiply by 10 when using the 100-ma. position, getting your readings from the 10-ma. scale. This is easy, however.

The greater the current range extension, the smaller the shunt resistor must be. With meters of low resistance, the shunt necessary to extend the range above 1 ampere is usually only a fraction of an ohm-so even a small amount of switch-contact resistance in series with the shunt might cause a large current to flow through the meter. avoid To this contact-resistance trouble on the high current range. many service instruments use the extra terminal C shown in Fig. 6. Terminals, A and B are used for the basic range and for the low extensions. For the high-current range, terminals B and Care used. Shunt R_3 is connected right between B and C. The selector switch is now in the meter circuit. In measuring high currents, switch-contact resistance will not matter greatly, as it adds to the meter resistance instead of to the shunt resistance. Also, the high current does not flow through the se-V. lector switch contacts. However, Fig. 6 does not represent the best arrangement, as it is possible to damage the meter if the switch develops too much contact resistance in position 2 or 3 or \checkmark should open-circuit when switching.

This damage is avoided in the circuit shown in Fig. 7, which is known as a "series or ring shunt." When the selector switch is in position 1, the lowest range is obtained. Note that this is not the basic meter range, as the resistors R_1 , R_2 and R_3 are acting as a shunt across the meter.

When the switch moves to position \mathcal{D} , resistors R_2 and R_3 are the shunt, while R_1 is added to the meter resistance, since it is now in series with the meter. Similarly, in position \mathcal{J} , R_3 is the shunt while R_1 and R_2 are in series with the meter.

There are two advantages to this arrangement; the meter is protected against damage at all times, and the resistors are relatively independent of the meter resistance, so they can be chosen to be any reasonable values.



FIG. 7. The use of a ring shunt makes this a better multi-range current meter.

Switch-contact resistance causes no trouble, since it adds no resistance to either the meter or shunt circuit. Should the switch become defective, the entire meter circuit is opened and protected instead of damaged.

USING A DIRECT CURRENT METER

There are five things you must keep in mind when using a current meter. Let's see what they are and why they're important. Then we'll summarize our findings in five easy-to-remember rules.

1. Series Connection. The most important thing to remember is that the current meter must be in series with the circuit. If necessary, unsolder a lead or cut a wire so the meter can be placed in the circuit. Remember, the circuit current you want to measure must pass through the meter itself. If a milliammeter is ever connected across a circuit part (in parallel), it is almost certain to be burned out.

2. Correct Point. In the simple circuit shown in Fig. 8, the meter may be placed anywhere in the circuit. The meter readings at A, B and C will be the same, because the same current flows through the entire circuit. However, when checking screen-grid or pentode tubes another problem arises. The screen-grid current also flows in the cathode lead in returning to the B supply. Hence, the plate current alone

can be measured only at positions B or C. Position A will give both screen and plate currents together.

When more than one tube is used. as shown in Fig. 9, notice there are now two paths for current, one for each tube. If you want to measure the plate current for tube VT_1 only, be careful to locate a point where only the plate current of this tube is flowing. The cathode lead of tube VT_2 may connect



FIG. 8. Be sure to place a current meter in series and use the proper polarity.

directly to the cathode of VT_1 and the B+ ends of resistors R_1 and R_3 may be directly connected together, so positions such as A and C of Fig. 8 may not be usable. The shaded areas of Fig. 9 show the proper places to measure the plate current of VT_1 . You must be sure to insert your meter where only the *desired* current is flowing. Otherwise, false readings will result. The meter may even be damaged if the sum of the currents exceeds the meter range you have chosen.

3. Current Range. The current range of the meter must be higher than the expected current; otherwise, the meter may be badly damaged or even ruined. The circuit current is determined by Ohm's Law, I = E/R, where R is the total of all the resistance in the circuit. In a circuit like Fig. 8, with a fixed supply voltage, the tube plate resistance plus resistor R_1 determines the current flow. The current will be about 5 to 10 ma. if this is an ordinary amplifier tube. However, if the tube or resistor R_1 is short-circuited in some way, the resistance of that part is removed and the current will be many times higher—perhaps enough to burn out the meter.

This danger is the reason that current measurements are not made very often in service work, except where voltage and resistance measurements indicate that it is safe to do so. When using a multi-range current meter, always start with the highest current range and switch to lower ranges if the readings permit. Voltage and resistance measurements properly made will usually make it possible to calculate the current, saving you the trouble of opening the circuit except in rare instances. (In transmitters it is necessary to know certain current values at all times for adjustment and maintenance purposes, so current meters are permanently connected in the proper plate and grid circuits.)

4. Meter Resistance. In the usual plate circuit, the tube and other parts



FIG. 9. The right meter position is important.

have resistances many times the meter resistance, so inserting the meter does not affect the amount of current flow much. For example, a meter of 100 ohms added into a circuit of 10,000 ohms obviously will not greatly change the circuit resistance. Suppose we were to apply 200 volts to a 10,000-ohm resistor. From Ohm's Law $(I = V \div R)$, the current would be .02 ampere or 20 ma. Now if the circuit resistance is in-



creased 100 ohms by the insertion of a 100-ohm meter, the total resistance will be 10,100 ohms. The current then will be .0198 ampere or 19.8 ma. You would not be able to read the average meter close enough to tell the difference between 19.8 ma. and 20 ma., so for all practical purposes, the meter has not changed the current flowing in the circuit.

However, if the meter is inserted in a low-resistance circuit, it will frequently change the current or may even determine the amount of current flow. For example, a tube filament may have a resistance of 40 ohms. If a 100ohm meter is connected in series, the original current will be cut considerably. If the voltage is 2 volts, the original filament current for a 40-ohm filament resistance would be 50 ma. With the meter resistance added in series, the total is now 140 ohms, so the current would be cut to about 14 ma., which is about 1/4 of the original value. Here you would have to depend on voltage and resistance measurements. In laboratory work particularly, the effect of the meter must be carefully considered, and the lowest resistance meter available, with the required range, must be used. If a suitable meter is not available, other types of measurements must be made.

5. Polarity. The direction of movement of the meter coil and the attached needle depends on the direction of the magnetic force, which in turn depends on the direction of current flow through the coil. If the meter is connected backwards, it will indicate backwards or down-scale.

The meter terminals are marked so that proper connections can be made. Just remember to connect the positive terminal of the meter toward the positive terminal of the source, so electrons *enter* the *negative* meter terminal.

To help fix this in your mind, refer

back to Fig. 8 where the correct meter polarity is indicated. When a tube is in the circuit, it acts as a signpost, showing the direction of electron flow through the entire circuit. Electrons, of course, flow only from the cathode of a tube to its plate and continue to flow in the same direction through the rest of the circuit. To make the meter needle read up-scale, the electrons must *enter* the *negative* meter terminal. Should the meter connections be reversed so the electrons enter the meter terminal marked +, the meter needle will read down-scale.

If the circuit does not contain a tube, you have to determine the polarity of the source and follow the rule of connecting the positive meter terminal toward the positive of the source.

Now let's summarize what we've learned about using current meters.

1. Always connect the meter in series, with the circuit.

2. In multiple circuits, connect the meter where only the desired current flows.

3. Always use a meter range higher than the expected current.

4. The meter (and shunt, if used) resistance must be *much smaller* than the circuit resistance.

5. Connect the meter so electrons enter the negative terminal.

PULSATING CURRENT

Suppose we tried to measure a.c with our d.c. meter. What would happen?

If it were very slowly changing a.c., the coil would turn first in one direction and then in the other, as indicated by the fact that we must watch polarity. This means that half of the time the meter needle would be off-scale. This would not happen with 60-cycle a.c., however, because the meter movement is too slow to follow a.c. variations occurring more than 5 or 10 times per second. For alternating currents

9 World Radio Histor of higher frequency than 5 cycles, the inertia of the meter movement makes the D'Arsonval meter indicate the *average* current over a period of time.

When you analyze an a.c. wave, you will notice that in each cycle the cur-



FIG. 10. The average value of a pulsating d.c. wave is equal to the d.c. component.

rent is alternately positive and negative, so the *average* current is zero. That is, the positive swings are equal to the negative swings, so their effect on a slow-moving object is zero. This is important; it means that a D'Arsonval meter alone will not read on a.c. it can only indicate the a.c. average (zero), regardless of the amount of a.c. current passing through it.

Now a pulsating current has both a d.c. and an a.c. component, as shown

in Fig. 10. Our meter won't read the a.c., but it will read the d.c. So, on pulsating d.c., a meter reading shows only the amount of d.c. present.

In the case shown in Fig. 11A, an a.c. cycle has been rectified by a half-wave rectifier. Now the pulses exist in the





same direction all the time, so this is another form of pulsating current. Even though there are gaps when no current flows, there is still an average value. (Naturally, this average is low; it's only about .3 of the peak value.) The D'Arsonval meter will indicate this average value. When a full-wave rectifier is used, the average value is higher, as shown in Fig. 11B. The average is now about .64 of the peak value.

Just remember this—any pulsating current has a d.c. average which can be read on a D'Arsonval meter.

Alternating Current Meters

In discussing alternating current meters, we will consider power line, a.f. and r.f. currents all as a.c. The only difference is one of frequency; however, this is important because some a.c. meters cannot measure the higher audio or r.f. frequencies accurately.

In the preceding section, you saw that a D'Arsonval meter could not by itself measure a.c., because the average value of a sine wave current over a period of time is zero. But our meter can be used to measure a.c. if we find some way to make this average value different from zero. Actually, there are three general classes of a.c. measuring devices, and D'Arsonval meters are used in two of them. The three classes are:

1. The D'Arsonval d.c. meter with a vacuum tube or copper-oxide rectifier.

2. The d.c. meter with a conversion device, such as the thermocouple or photoelectric cell.

3. True a.c. meters, such as the hotwire, magnetic-vane and the electro-

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dynamometer types.

Let's examine each type to see how it works.

ALTERNATING CURRENT RECTIFIERS

Vacuum Tubes. Any diode tube will rectify a.c., as current can pass through the tube only when the plate is positive with respect to the cathode of the tube. In the circuit shown in Fig. 12, the diode tube VT passes current only when terminal 1 is positive with respect to terminal 2. The resulting half-wave pulses have an average value which is about .3 of the a.c. peak. This average value will produce a deflection on a D'Arsonval meter.



FIG. 12. A diode can be used to rectify a.c.

The condenser C in Fig. 12 is used to by-pass the high-frequency pulsations around the meter. The meter movement is a coil and so has inductance. If not by-passed, the inductive reactance would be added to the meter resistance and might both upset the circuit and make it non-linear with respect to frequency. The condenser also serves to smooth out the pulsating current, bringing the peak value down and raising the d.c. average. This not only gives a greater sensitivity, but also reduces the danger of high peak values damaging the meter.

The rectifier tube and meter can't be placed in series with the circuit whose current we wish to measure, because the tube will change the a.c. in the circuit to a pulsating current. We want rectification for the meter, not for the complete circuit. We get it by inserting resistor R in series with the circuit. The a.c. voltage developed across the resistor then is applied to the tubemeter combination. Thus, we are actually measuring the current indirectly by the voltage drop produced in this resistor. The meter is calibrated so all these conditions can be accounted for. Usually, it indicates the effective or r.m.s. value of the a.c.

Figure 13 shows how a triode tube may be used for rectification. The C bias is adjusted so that the plate current is nearly zero. When an alternating current flows through resistor R, an a.c. voltage is developed across it. Negative alternations make the grid even more negative and so have practically no effect on plate current. However, the positive alternations swing the grid in the other direction so that plate current pulses flow. The meter then responds to the average plate current.

This plate current is proportional to the a.c. grid voltage, which is determined by the alternating current in the circuit and the value of resistor R. Thus, the plate current is an indirect measure of the circuit current.



FIG. 13. How a triode rectifier is connected.

Resistor R acts just like a shunt in this circuit. For example, we may need 1 volt on the grid to produce a fullscale meter deflection. By using a 1000-ohm resistor, a circuit current of 1 ma. would give 1 volt. If we use a 100-ohm resistor, 10 ma. would be needed to produce 1 volt.

Copper - Oxide Rectifiers. The D'Arsonval meter and copper-oxide rectifier* combination is the most

^{*}Selenium on iron is also used in place of copper oxide, and the same rectifier circuits are used.

widely used a.c. measuring device, and is found in the equipment used by most servicemen. The copper-oxide rectifier is stable, relatively reasonable in cost, and can be designed to work on small currents.

Each rectifier element consists of a copper disc or washer, which has one surface oxidized by a special heat treatment. A contacting washer, usually made of lead, is pressed on the oxide layer. This oxide layer rectifies because it permits electron flow through the layer in one direction much more readily than in the other. In other words, electrons can move easily from the copper through the oxide to the lead contacting disc but they encounter a fairly high opposition when trying to move in the other direction.

There are several different ways of connecting these elements, but the most common is the full-wave bridge circuit shown in Fig. 14. Four separate elements are used in the bridge. In practice, these elements would be mounted on an insulated single bolt, with insulating washers between the elements.

The arrows alongside the elements in Fig. 14 indicate the directions in which electron movement most easily occurs for each half cycle. When terminal 1 is negative (2 is positive), the electrons move in the direction shown by the solid arrows. At point A the electrons choose the path toward point B, because rectifier W conducts better than rectifier X for this half of the a.c. cycle. At point B, the electrons go through the meter instead of rectifier Z, since Z offers a high impedance to electrons moving in this direction. Reaching C, the electrons pass through rectifier Y to point D, and from there go to terminal 2.

When the cycle reverses, terminal \hat{z} becomes the negative terminal. Now electrons follow the dotted arrows from terminal \hat{z} to D, then through rectifier

Z to B. From here, the path is through the meter to C, then through rectifier X to A and so to terminal 1. Note that regardless of the polarity at 1 and 2, electrons always pass through the meter in the same direction, which is exactly the action of a direct current. Hence, a direct current meter can-be used at M.

We use full-wave rectification in this circuit because it gives a higher



FIG. 14. The copper-oxide rectifier in a bridge is the most common rectifier for service work.

average current, about twice that for half-wave rectification. This practically doubles the meter sensitivity by giving a higher meter deflection for the_____ same amount of alternating current, so that our meter has an a.c. sensitivity almost as good as that on d.c.

If one element of the bridge burns out, half-wave rectification will be obtained instead of full-wave. Then, since the average current is only about half that for full-wave, the meter will only read about half as much as it should. You can easily spot a damaged rectifier element in a meter of this type —check a known a.c. current; if the reading is half what it should be, one of the elements is damaged.

The back-to-back circuit is another form of full-wave bridge, with the advantage of using only two rectifier elements instead of four. The circuit gets its name from the rectifier connections, where the "backs" are connected together as shown in Fig. 16. The resistors R_1 and R_2 replace two of the rectifier elements of the usual bridge circuit, and are chosen to have a resistance about halfway between the conductive and non-conductive values of a copper-oxide element. (An element may have only 100 ohms opposition to electron flow one way and several thousand ohms opposition to electron flow in the other direction.)

There is a continuous a.c. path through the resistors, but this current does not pass through the meter. When electrons move from source terminal 1. rectifier W becomes conductive, so there will be an electron flow through this rectifier to point B. Rectifier Z opposes current flow by having more resistance than the path through meter M and resistor R_2 , so the major electron movement is from 1 to A, through W to B, through M to C, through R_2 to D and thus to 2. On the next half cycle, the major electron movement is from terminal 2, through rectifier Z, then through the meter and resistor R_1 back to terminal 1. Thus, when the rectifier is conducting, it offers less opposition than the resistor directly across from it, but when the cycle reverses the resistor is more conductive than the rectifier.

To extend the a.c. range with either form of bridge, shunts are connected between terminals 1 and 2 instead of across the meter. If shunts were used across the meter, higher currents would have to flow through the rectifier elements and would burn them out.

Copper-oxide rectifiers have two faults: 1, they are easily damaged; and 2, they have a high shunting capacitance between terminals.

Reasonable care to prevent overloading will protect them from burnout. They must not be exposed to high temperatures for very long. Heat tends to change the chemical characteristics of the oxide layer, causing a change in the rectifying ability of the unit. Too much heat, such as from a soldering iron, can ruin a rectifier.

The high self-capacitance generally limits the copper-oxide rectifier to power line and the lower audio frequencies. By reducing the size of the rectifier elements the capacity is reduced, so the frequency range can be extended somewhat.

Of course, the capacitance does not prevent the use of these rectifier-meter units as r.f. current indicators; it just changes the calibration. With increases in frequency, the capacity acts as a decreasing shunt reactance, so the current range changes rapidly. However, reasonably accurate readings on the original meter scales of some types of meters can be made up to 80 kc. without special calibration charts. Since you, as a serviceman, will be primarily interested in measuring operating volt-



FIG. 15. The back-to-back bridge saves two rectifier elements.

ages and currents which are d.c., power line a.c. or low audio frequency a.c., you can usually ignore the frequency characteristics.

CONVERSION DEVICES

Thermocouple. The need for a sensitive indicator for r.f. currents led to the development of the thermocouple, commonly found in transmitter installations and research laboratories.

This device works on an interesting principle—the fact that when a junction of two dissimilar metals is heated, a d.c. voltage is produced. Thus if a junction between a copper rod and an iron rod is heated in a flame, a d.c. voltage is produced; the two unjoined ends of the rods are now the d.c. terminals. This method of getting a voltage suggests a means of converting a.c. power through its heating effects.

We know that current flow through a resistance produces heat, so if we can bring a junction of dissimilar metals up to a resistor acting as the source of heat, the d.c. voltage so developed can be used to operate a d.c. meter. Figure 16 is a typical arrangement.



FIG. 16. The thermocouple is widely used for r.f. measurements.

The wire between terminals 1 and 2 is a resistance wire, called the heater. The wires between A and 3 and A and 4are the thermocouple elements.

The heater wire is made short, and the ends are connected to large terminals so that heat developed near the ends will be rapidly conducted away. Hence, the heat is concentrated at point A, the center of the heater. The dissimilar metal junction, called the thermocouple, is placed at A. In some cases, the thermocouple is just near the heater, while in others it is welded to it. Since the voltage produced for a particular temperature depends on the metals used in the thermocouple, these metals are carefully chosen to give the greatest d.c. voltage. The wire between 3 and A may be platinum, while the wire between 4 and A may be constantan.

The current to be measured is per-

mitted to flow through the heater wire, between terminals 1 and 2. The resulting heat is proportional to the power dissipated in the resistance of the heater, which in turn makes the heat proportional to the square of the current. $(P = I^{\circ}R)$ The comporature_ rise heats the thermocouple junction at A, producing a d.c. voltage which will cause a d.c. current flow through the meter. Notice this is a produced d.c. and is not the current being measured, which is usually a.c.

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The heating effect of a current flow through a resistance will occur regardless of frequency or wave form, so any a.f. or r.f. current can be measured by this device. In fact, the thermocouple is practically a standard r.f. current meter. To eliminate skin effect^{*}, the resistance wire is made in the form of a hollow tube in models intended for ultra-high frequencies.

These instruments are made in many ranges, from a few milliamperes up to practically any value wanted. An instrument sensitive to low currents is made by sealing the thermocouple and heater in a glass tube from which the air has been pumped. This is known as a vacuum thermocouple.

The use of a thermocouple has another advantage—the meter and thermocouple do not have to be mounted together. This is important, as currents frequently must be measured in hardto-reach places, such as in a transmission line, transmitter tank circuit or antenna installation. It is impossible to run r.f. currents over long leads to a more convenient meter location without upsetting the circuit. Therefore, the thermocouple and heater unit is

^{*} Skin effect is the tendency of high-frequency currents to flow on the surface instead of through the center of the wire. This produces the effect of a change in resistance. By eliminating the center of the wire, all current must flow along the surface, so there is no change.

placed right in the circuit containing the current to be measured. Then the d.c. produced is fed over a feeder line to the meter, as shown in Fig. 17. Choke coils L_1 and L_2 , together with condensers C_1 and C_2 , filter r.f. currents out of the line to the meter. Should the thermocouple be connected in a circuit where one terminal of the heater is grounded, the filter probably would not be required. If the line to the



FIG. 17. The thermocouple permits remote meter readings.

meter passes through any r.f. fields, it may pick up some r.f. energy which could cause an undesired current through the meter. Another condenser (or a pair connected like C_1 and C_2) may be connected across the meter terminals to by-pass this r.f.

The meter is calibrated by passing known currents through the heater element. These currents are usually d.c. or low-frequency a.c., which can be more accurately determined than r.f.

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As the deflection is proportional to the square of the a.c. being measured, the meter scale will be crowded at the lower end if a standard meter is used. That is, the deflection is smaller than normal for small currents, and is greater than normal for high currents. By using a meter with specially shaped pole pieces, it is possible to obtain a progressively decreasing d.c. sensitivity as the meter moves up-scale, so that the scale is more uniform.

Such a meter is shown in Fig. 18. The pole faces have a shape which creates a non-uniform field between them with the maximum lines between the closest points (1 and 2). When the

meter coil is in the position shown in Fig. 18, the twisting force caused by the reaction of its field with the meter field is a maximum, so the coil will give a large deflection as the current through it increases. The more the coil rotates to the right or clockwise, the smaller this twisting force becomes, so larger coil currents are needed to give the same amount of deflection. Thus we have the decreasing sensitivity at the upper end of the scale that we were looking for, and our scale becomes practically uniform over its whole length.

This special method of cutting the meter pole pieces increases the meter cost, but it is a means of getting an extended current range without sacrificing the ability to read small values accurately.

Photoelectric Types. This method is used only in experimental work, but you may meet it, so we'll go into it briefly.



FIG. 18. Using pole pieces having a special shape produces a non-uniform magnetic field.

The amount of light obtained from an electric light bulb varies according to the amount of current flow through the bulb. By connecting a photovoltaic cell to a meter and exposing the cell to a light bulb, the light will produce currents through the cell and meter proportional to that through the bulb. Then we can calibrate the meter in terms of the current flow through the bulb. The bulb can then be connected in the circuit carrying the current, and the photocell-meter used to indicate the current. This method is not very accurate—it's more useful to indicate when changes occur or when maximum current flows than it is to indicate the exact amount of current.

TRUE A.C. METER MOVEMENTS

You recall the D'Arsonval meter cannot be used directly on a.c. because it cannot follow the rapid reversals of current. Even if it could, it would reverse its direction when the a.c. reverses. This is because the coil field 19A. When current flows through the coil, the vane is magnetized by the coil's magnetic field, with a magnetic polarity opposite to that of the coil. Since opposite poles attract, the vane will be attracted into the coil until the magnetic force is balanced by-thespring. The vane is soft iron, so it does not become a permanent magnet. Hence, when the current flow through the coil reverses, the reversal of the coil magnetic field also reverses the polarity of the magnetic vane and the attraction is maintained.

Figure 19B shows another type of magnetic-vane meter. Two pieces of



FIG. 19. Here are two common types of magnetic vane movements.

reverses with the a.c. flowing through it, but the fixed magnetic field continues to have the same polarity. If we can arrange the meter so both magnetic fields reverse at the same time the a.c. reverses, the magnetic turning force will always be in the same direction and the meter will indicate just as if d.c. were applied. There are two meters that do this, the magnetic-vane and the electrodynamometer types.

Magnetic-Vane Meters. This meter works in a different manner from the D'Arsonval movement. Instead of using a fixed magnet and a moving coil, the coil is fixed and a soft iron vane is the moving element, as shown in Fig. soft iron are used in this type, with the coil wound around both. The triangular piece AB is bent in a cylindrical shape and is mounted on the inside of the coil form. The rectangular piece CD is also bent in a cylindrical shape. This piece is connected to the spindle XY, to which is attached the meter pointer and restoring spring.

When current flows through the coil, both pieces of iron are magnetized. As the coil field is along the axis of XY, the top and bottom edges of vanes AB and CD will become the magnetic poles. The edges C and D will have the same polarity as A and B respectively. Thus, C and A may be north poles, while D and B may be south poles. This means like poles are near each other and will repel. The soft iron pieces are so placed and shaped that CD rotates to the right (clockwise). Again, the movement will, be to the point where the spring balances the magnetic force.

When the coil field reverses, both magnetic poles reverse, so the magnetic opposition remains the same as before.

Although several other types of magnetic-vane movements have been developed, they all have similar characteristics. To produce the strong magnetic field required, the coil must have a large number of turns. The coil thus has considerable inductance, which will upset the calibration if high-frequency currents are measured. Also, there is a limit to the frequency with which the vane magnetism can be changed, so these meters are used only for d.c. and low-frequency a.c. measurements

The inductance of the meter coil makes it hard to use shunts for extending the range, because the coil reactance changes with frequency — so these meters usually have only one range. In spite of all its shortcomings, low cost and rugged construction have made the magnetic-vane meter very popular for use where great accuracy is not required.

Electrodynamometer. This meter is a moving-coil type, but is quite different from the D'Arsonval meter. No fixed magnet is used; instead, the field is produced by an electromagnet, and the moving coil has an air core. Figure 20 shows a typical electrodynamometer. Coil A is the field coil, which may be wound on any suitable coil form. The inner coil B is the moving coil, made so it can rotate within coil A. The two coils are connected in series. When current flows through the coils, both develop magnetic fields proportional to the current strength and the number of turns on each coil. This. causes coil B to rotate until the magnetic force is balanced by the spring, as in other meters.

When alternating current is measured, the reversal of the direction of current reverses both magnetic fields, so the repulsion remains the same as before. Hence, the meter can be used either for a.c. or d.c.

The moving coil has a tendency to waver back and forth a while before



FIG 20. This electrodynamometer does not use a permanent magnet.

settling down to the correct reading, just as with other meter movements. This wavering is damped by a vane attached to the moving coil and fixed to rotate through a closed chamber known as a damper box. When the coil moves, the damper vane also moves, compressing the air ahead of it in the damper box. The compressed air acts as a cushion, so that the pointer comes slowly up to the correct reading and then stops.

Because of the inductive effects of the coil windings, the frequency range of such meters is limited. The usual types hold their calibration for frequencies up to 130 cycles; special meters can have ranges up to 1000 cycles or more, but will be accurate only for frequencies very close to the frequency for which they were built.

The original electrodynamometers were used as standard instruments, as they could be calibrated directly from their dimensions. Modern commercial electrodynamometers are calibrated by comparison with a standard, however, just like other meters.

As the same current flows through both of the series-connected coils and as the deflection depends on the strength of both magnetic fields, the deflection is proportional to the square of the current for direct current. For an alternating current, the meter deflection is proportional to the square of the effective or r.m.s. value. Since an alternating current of any r.m.s. value will produce a magnetic field equivalent to that produced by a direct current of the same value, the meter scale is exactly the same both for d.c. and low-frequency a.c. This makes it possible to compare the effective values of currents regardless of wave form.

Large currents cannot be measured easily with an electrodynamometer, because the moving coil cannot be made of exceedingly large wire or have heavy leads going to it. Although shunts can be used to a certain extent, a current transformer is the most common way of extending the range of this meter.

Current Transformers. As you will recall, a transformer will step up or down an a.c. voltage according to the ratio of turns on the primary and secondary windings. Also, neglecting losses in the windings and core, the primary power will equal the secondary power. This means that if the secondary voltage is stepped up, the secondary current will be reduced in the same ratio so that the power $(I \times E)$ will be the same. Hence, a transformer which steps up voltage by a ratio of 1 to 3 also steps down current by a ratio of 3 to 1. For example, if the primary voltage and current are 110 volts and 4 amperes, the primary wattage is 440 watts. If the secondary voltage is stepped up three times, to 330 volts, the secondary current must be one-third the primary current (1-1/3 amps.), so that the secondary power will also equal 440 watts.

Therefore, by designing a step-up voltage transformer with low losses and a small primary, we have a stepdown current transformer in which a large primary current flow will result in a small secondary current which canbe indicated on a meter. This secondary current is proportional to the primary current, so the meter can be calibrated in terms of primary current.

Current transformers are usually found where very large currents are



Courtesy Weston

FIG. 21. Current transformers. The core of the one at the left is snapped around the currentcarrying wire. The one at the right is installed permanently, passing the power line through the center. These transformers are usually used on high-current power lines, where currents of 25 amperes to thousands of amperes are to be measured.

measured. Where the current is several thousand amperes, the current transformer may be just clamped around the conductor carrying the current, allowing this conductor to act as a singleturn primary winding. Typical current transformers are shown in Fig. 21.

Although commonly used with electrodynamometers, current transformers can be used with any other a.c. meter having the required characteristics, such as the proper range and proper impedance. However, since the type of meter used may have an effect on the calibration, the current transformer and meter combination have to be calibrated as a unit. Hot-Wire Meter. Another method which has been used to measure current depends on the fact that metals expand when heated. Any kind of current (a.c. r.f. a.f. or d.c.) with any wave form will heat and expand a resistance wire through which it passes; the amount of expansion will be a measure of the heat produced, and therefore of the current flowing. We can use this principle in a meter by making the expansion cause movement of a needle.

Figure 22 shows a method of doing this. The wire which is heated is shown between A and B. This wire is held under tension by an adjustable spring (not shown). Attached to this wire at C is a metallic cord D which passes around the pulley P, continuing to the spring S which is fastened at E. The pulley and support E are made of insulating materials to prevent grounding the supply circuit through the meter frame or case. When a current flows through the wire A-B, it heats and stretches, sagging downward. This allows the spring S to pull the cord Daround the pulley P, thus rotating the pulley to the right. The meter pointer X is attached to the pulley and is thus moved over a calibrated scale.

This meter measures only high cur-



FIG. 22. The mechanism of a hot-wire ammeter.

rents and is easily burned out. Elaborate means of compensating for surrounding temperature changes must be used. Also the amount the wire stretches when a given current passes through it changes with repeated use, so such meters are no longer very popular.

Voltmeters

Voltages, like currents, are measured by meters. In fact, a current meter can be used for voltage measurements if properly calibrated. You will recall that a certain voltage is necessary to force the required current through the resistance of the current meter. We'd be justified in saying that meter deflection is proportional to the applied voltage, and in calibrating our meter accordingly.

As an example, a 1-milliampere meter may have a resistance of 200 ohms. The voltage required for full-scale deflection is $E - I \times R$, or $.001 \times 200$ - .2 volt. Hence, we can use this meter as a 0-.2 voltmeter. This range is too low for ordinary service purposes, but we can increase it by increasing the meter resistance.

Since increasing the resistance of the meter movement itself is not practical, resistance is added in *series* with the movement, so that a higher applied voltage is needed to drive full-scale current through the meter. If 800 ohms external resistance were added to the 1-ma. meter in the example above, 1 volt would be needed to give a fullscale reading.

By adding sufficient resistance, any desired voltage range can be obtained.

As with current meters, a single meter having several ranges is desirable, so a low basic range is usually chosen and then additional resistors are connected in series to extend the range further. A typical switching arrangement is shown in Fig. 23. Resistor R_1 , together with the meter resistance, sets the basic voltage range. Then other resistors are added by the switch to obtain the desired extensions.

Notice that a current flow through the meter is being interpreted as voltage because of the known resistance of the meter. This current must come from the source being measured, in addition to that taken by the load, because the voltmeter is connected *across*



FIG. 23. Voltmeter ranges are extended by using multiplying resistors in series.

the terminals where the electrical pressure (voltage) is to be measured.

When the source has a low internal resistance (as is the case with a good battery) and the source is not supplying appreciable current to a load, the current taken by the voltmeter scarcely matters. However, if there is appreciable resistance in the source or between the point of voltmeter connection and the source, the readings will vary with the sensitivity of the voltmeter. Let's learn more about this now.

VOLTMETER SENSITIVITY

As a practical example of a typical radio circuit, consider Fig. 24. Suppose that resistor R_1 has a value of 50,000 ohms and the tube plate-cathode resistance is 25,000 ohms. The total resistance is then 75,000 ohms, so 4 ma. flows from the 300-volt source $(300 \div 75,000 - .004$ ampere). The tube plate-to-cathode voltage will be 100 volts, and the drop in resistor Rwill be 200 volts.

Now suppose we connect a voltmeter as shown. Will it indicate 100 volts? This depends on the current taken by the meter, which must flow through resistor R_1 , in addition to the plate current of the tube. In other words, the meter is in parallel with the tube resistance, so our circuit is really that shown in Fig. 25. A few practical cases will help you see what happens.

Case I. If the voltmeter in Fig. 24 has a 100-volt range which requires 10 ma. for full-scale deflection, its resistance $R_{\rm M} = E \div I$, or 10,000 ohms. This resistance is in parallel with the tube resistance $R_{\rm P}$ of 25,000 ohms, as shown in Fig. 25, so the resulting parallel combination of meter and tube is

 $R_{\rm P} + R_{\rm M}$ 25,000 + 10,0007100 ohms, approximately. This value is in series with the 50,000-ohm resistor R_1 , so the total circuit resistance has changed from 75,000 ohms to about 57,100 ohms. The new total current from the 300-volt source is about 5.25 ma. $(300 \div 57,100 - .00525 \text{ ampere})$. This current through R_1 gives a voltage drop of 262.5 volts, leaving only 37.5 volts for the tube and meter parallel combination. In other words, the tube plate voltage is now only about 7/57 of the total, instead of 1/3. (The ratio is 7100 to 57,100 instead of 25,000 to 75,000.) This means that the voltage measured by the meter is about 38 volts, instead of 100 volts. The connection of the voltmeter has upset the voltages so that more drop exists across R_1 (now about 262 volts) and less across the tube, giving a false indication of the conditions existing without the meter.

Case 2. Suppose we now use a voltmeter having a 100-volt range which only requires 1 ma. for full-scale deflection. The meter resistance is now 100,000 ohms. Figuring current and voltages as we did in Case 1, we find that the plate voltage now measures 85 volts, much closer to the original 100 volts.



FIG. 24. A typical voltage-measuring problem. The accuracy depends on the ohms-per-volt sensitivity of the meter.

Case 3. A 100-volt meter requiring only 0.1 ma. has a resistance of 1,000,-000 ohms. Such a meter will indicate about 98 volts if placed in the circuit of Fig. 24. The error now is only about 2%, well within the normal tolerances allowed for this kind of circuit.

Obviously, in any circuit having appreciable series resistance, the lower the current needed to give deflection, the more nearly correct the results.



FIG. 25. Notice how the voltmeter resistance is in parallel with the tube plate-cathode resistance.

For this reason, voltmeters used in radio service work require 1 ma. or less for full-scale deflection.

Ohms-Per-Volt. The term "ohmsper-volt" is used to express voltmeter sensitivity. The smaller the current needed for full-scale deflection, the greater the sensitivity of the meter. Also, the smaller the current for a given voltage, the greater the meter resistance must be. $(R = E \div I)$ By dividing the total meter resistance by the basic full-scale voltage range, the ohms-per-volt rating will be obtained. Thus a 2-volt meter having 2000 ohms resistance is a 1000 ohms-per-volt meter. Ohm's Law will let you figure ohms-per-volt ratings easily. Notice that this rating increases as meter sensitivity increases.

Now here's something you should remember. Service manuals usually give you radio circuit voltages (plate-tocathode voltages, etc.) which are taken with 1000 ohms-per-volt voltmeters. But if you are using a more sensitive meter — say 20,000 ohms-per-volt the readings you get when you measure these circuit voltages will be higher than the manual says for any highresistance circuits, but will be more nearly correct for the actual circuit operating conditions. (You just learned why this happens.) Just remember this -set voltages will be *higher* than their service manual ratings in resistance coupled circuits, if you measure them with a voltmeter having a sensitivity greater than 1000 ohms-per-volt. How much higher, of course, depends upon the sensitivity of the meter, and the resistance in the circuit where it is connected.

Knowing the ohms-per-volt rating of the meter makes it easy to extend the range. Just multiply the ohmsper-volt rating by the number of volts the range is to be extended. For example, if a 1000 ohms-per-volt meter is to be extended from 100 volts to 500 volts, the extension is 400 volts. Multiplying 400 by 1000 gives 400,000 ohms as the needed extra series resistance.

A.C. VOLTMETERS

Any of the standard alternating current meters can be used as an a.c. voltmeter in exactly the same manner

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just described. For service work. the copper-oxide rectifier and D'Arsonval meter combination is generally used. This type has better sensitivity characteristics than any except the vacuum tube voltmeter, which will be taken up in another lesson.

The copper-oxide rectifier type a.c. voltmeter usually has a sensitivity of 1000 ohms-per-volt, requiring 1 ma. for full-scale deflection. Even lower sensitivities are permissible in service work, where the usual a.c. meter readings are filament voltage, power line voltage, rectifier plate voltage and output voltages during alignment. The sources for these first three measurements have low series resistance, so the current drawn by the meter does not greatly matter. For output readings, only an indication of the maximum voltage, not the exact amount, is usually desired.

Of course, the copper-oxide unit has the same faults described before; it is easily overloaded and has shunting capacity. When overloaded, one element usually burns out so readings fall to half-normal. The effect of the capacity of the copper-oxide rectifier is more pronounced with a voltmeter than with a milliammeter, because of the series resistances. However, the copper-oxide types are reliable up to 1000 cycles or so, which is sufficient for output purposes, as most signal generators are modulated at about 400 cycles.

Ohmmeters

Since we can now measure voltages and currents, we can use Ohm's Law and calculate the resistance in a circuit or device. However, this is a cumbersome method; a more speedy way of finding the resistance is desirable.

Suppose we make up a circuit consisting of a known fixed voltage source in series with a current meter. When we connect a resistor into this circuit, the amount of current flow will be inversely proportional to the value of the resistor. Then we can calibrate the meter in terms of resistance, and we will have a resistance indicator or ohmmeter.

Figure 26 shows how such a meter is designed. The resistance R_2 is adjusted so that its resistance plus that of R_1 and the meter is just sufficient to permit a full-scale deflection on the current meter (using a known battery voltage) when the test prods are touched together. Then when the circuit is opened and the unknown resistance R_x is inserted, the meter will indicate some lower value, depending on the added resistance.

As the battery voltage will decrease with age and use, and also since the meter and resistance values may be off somewhat, resistor R_2 is adjustable. Actually, R_1 and R_2 could be a single resistor, but if adjustments are not correctly made, the meter could be damaged. Hence, R_1 is fixed and has a value sufficient to prevent meter damage.

This adjustable resistor is set for a full-scale meter deflection, which adjusts for zero ohms on the scale.

As the battery ages, its voltage drops, so R_2 is decreased to restore zero. This change in voltage and resistance upsets the calibration, however; the ohmmeter accuracy will vary directly with the percentage change of battery voltage. Thus, a 10% drop in battery voltage will cause a 10% error in the ohmmeter reading. Fortunately, battery voltages are relatively stable over long periods of time, so replacements are not too frequently required.

By designing the ohmmeter for the rated battery voltage, advantage can be taken of the fact that battery voltages are somewhat higher when new. Then the ohmmeter will read a few per cent high with fresh batteries, and will then drop to exact readings and gradually read lower as battery voltage decreases. This has the effect of halving the error, as a 10% battery voltage change now produces a 5% variation each way from the correct reading. The sizes of R_1 and R_2 are usually chosen so that when R_2 can no longer be adjusted for zero ohms, the percentage of error has exceeded that set by the manufacturer, so a new battery must be obtained.

The ohmmeter is definitely a service instrument—it does not have sufficient accuracy for laboratory work. However, in service work, parts may normally vary as much as 20% anyway,



FIG. 26. This series ohmmeter is best for medium and high-resistance values.

so an error of a few per cent in measurement is not of great importance.

The fact that voltage variations will upset the ohmmeter is one of the reasons that power packs are not more widely used with ohmmeters. Only a well-regulated power pack will have a sufficiently stable output voltage.

The type of ohmmeter we have been considering is known as the series type, as the battery, meter and resistances are in series. This type can easily be recognized, as the "zero-ohms" position is at the right of the meter scale, with increasing resistance values to the left. This is just the reverse of the ordinary meter scale.

As with other instruments, the range is of importance. In service work, the wide range of resistances met (from .05 ohm for r.f. transformers to sev-



FIG. 27. A typical series ohmmeter scale having a 3000-ohm center-scale value.

eral thousand megohms for leakage measurements) presents quite a problem. Let's investigate this further.

Useful Range. Having made an ohmmeter using some particular meter and total resistance, just what range of values will it indicate? Let us suppose we are using a 1-ma. meter and a 3-volt battery. This means the meter $+ R_1 + R_2$ resistance is adjusted to 3000 ohms so 1 ma. can flow. You will see that the meter will read mid-scale (half the current) when a resistance of 3000 ohms is being measured, as we then have doubled the circuit resistance and still have the same voltage. Now, the most useful portion of the meter scale will be between values about 100 times the mid-scale value and 1/100 the mid-scale value. Thus, if the midscale value is 3000 ohms, the ohmmeter will be best calibrated and easiest read between 300,000 ohms and about 50 ohms. Higher and lower values can be read on this scale, but the readings become hard to make. Actually, this range is probably readable up to about 500,000 ohms, and values of 10 ohms or so may be estimated.

Figure 27 shows a typical ohmmeter scale based on these values. The

crowding at the low end is caused by the fact that the meter resistance is becoming much larger than the resistance being measured, so that changes in the latter have less and less effect on the current flow.

At the other end of the scale, the current is primarily determined by the external resistance. When the external resistance gets large, adding more resistance doesn't change the current much. Thus, at the low-resistance end, a scale division may represent only 50 ohms, while at the other end, several thousand ohms may be indicated by the same space on the scale.

The series ohmmeter is excellent for medium and high ranges, but is not the best type for low ohmic values. For measuring low resistances, we can lower the series resistance, but the lowest normal battery voltage is 1.5 volts, so there must be enough resistance to limit the current at the fullscale value. By reducing the meter resistance through the use of high current meters or shunts, or by using voltage dividing circuits to get lower voltages, somewhat better results can be obtained. However, this requires complex circuits and means a high current flow, which in turn means short battery life and possible damage to the device being tested.

Low-Range Ohnmeter. A shunttype ohnmeter circuit, shown in Fig. 28, is used to measure low values of resistance. The resistor R is adjusted until the meter reads full-scale with the test leads separated. When the test leads are connected to the resistance being measured, a lower meter reading results, as the unknown resistance R_x is connected in parallel with the meter and acts as a shunt across the meter. Notice these two important facts: (1) zero ohms on the shunt meter is at the left, exactly opposite to a series ohmmeter; (2) if the test probes are held together, the meter is short-circuited and cannot read. You will see that the "zero adjustment" on a shunt meter is actually the full-scale adjustment.

This circuit is excellent for low \overline{re} sistances. By using a low-resistance meter, the battery current flow will be primarily determined by resistance R, which can be fairly high. For example, if a 27-ohm, 1-ma. meter were used with a 1.5-volt battery, resistor R would have to be adjusted to about 1473 ohms to limit the current to 1 ma.



FIG. 28. This shunt ohmmeter is excellent for low-resistance values.

Since the resistance of R is so high compared to that of the meter, shunting the latter with even a very low-resistance device will have very little effect on the current drawn from the battery. We can assume the current flow is a constant amount and just divides between the meter and the unknown R_x .

A resistance equal to the meter resistance gives mid-scale deflection (since the current then divides equally between the meter and the unknown resistor). The best scale range is again from about 100 times to 1/100 the center scale value.

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High-Range Ohmmeters. As mentioned before, the series ohmmeter is a medium and high-resistance indicator. However, there is a practical limit to the range, which is determined by the current required for the meter. Until recent years, service instruments were usually 1-ma. meters—and such meters require very large power supplies to measure very high resistances. For example, using a 1-ma. meter to `obtain a useful scale above 50 megohms requires 500 volts or so.

Servicemen were thus limited to ranges readable up to 1 or 2 megohms. However, when more sensitive meters became generally available, one of the first results was an extension of the upper ohmmeter range. Meters with 50 to 200-microampere ranges (.05 to .2 ma.) permit the construction of ohmmeters of 20 to 50-megohm ranges, using no more than 45 volts.

Even these meters aren't sensitive enough to measure small amounts of leakage in condensers or in shielded cables. Laboratories formerly used "meggers" for this purpose. These megohmmeters were high-voltage devices with a hand-operated 500-volt generator and a special meter movement.

In recent years, vacuum tube circuits have been developed which measure values up to 1000 megohms and higher. These have replaced "meggers" generally; they will be taken up elsewhere in your Course.

Using Ohmmeters. Since the ohmmeter depends on its own battery for operating power, it is not used like a voltmeter or current meter. The latter two measure operating conditions, so the radio circuit being checked must be connected to its normal source of power and must be turned on. On the other hand, the circuit *must be turned* off when an ohmmeter is being used. This is an advantage at times, as the circuit need not be in operating condition when an ohmmeter is being used.

When making tests for circuit continuity, the ohmmeter is connected between selected points, as you will learn in other lessons. Then, when the defective circuit has been located, the various parts are individually checked. When a certain part is to be checked, remember that any parallel path in the set can give false readings. Hence, the part being checked should be disconnected from other parts if the results appear unusual.

The shunt-type meter draws current continuously from its battery as long as it is connected. Therefore, it always has a switch or other means for disconnecting the battery when readings are not being made. Remember to turn off this meter when not in use; otherwise, the battery will run down quickly.

Multimeters

As you have probably noticed, the same kind of meter can be used for voltage, current and resistance measurements. The only differences are the ways in which resistances are arranged as shunts and multipliers, the use of rectifiers for a.c., and the variable resistances and battery used on ohmmeters. Naturally, it would be very handy to have all these functions combined in a circuit using a single meter. That's exactly what the modern service multimeter is.

There are many types of multimeters, having the same basic functions but differing in appearance, method of switching ranges, meter sensitivity and sometimes in the number and extent of the ranges available. These can be separated into two types, those using rotating switches for selecting ranges and functions, and those using plug-in or pin jacks for this purpose. Let's consider the latter type first.

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PIN-JACK TYPE

Figure 29 shows a typical circuit. There are five controls on this instrument—two switches and three ohmmeter adjustments. Switch SW_1 is used to set the meter for a.c. or d.c. measurements.

There are eighteen jacks on the tester into which the test leads can be plugged. By choosing the proper pair of jacks, the proper range is obtained automatically. When a different range is needed, another pair of jacks is used.

Resistors R_1 , R_2 and R_3 serve as a series shunt across the meter for d.c. current measurements. The jack tifier in the circuit. The jack marked " $AC\pm$ " is the common a.c. voltage jack so one test lead is plugged in here for a.c. readings. However, at times you have to make a measurement in a circuit containing both a.c. and d.c. As the d.c. would cause false a.c. readings, it is blocked out by a condenser C, connected to a jack marked "output." This jack is used as the common Jack whenever the a.c. only is to be read. The other test lead is plugged into the "A.C. VOLTS" jack corresponding to the proper range. Resistors R_{s} to R_{11} are the necessary voltage multipliers. Resistor R_{12}



FIG. 29. A multimeter of the pin-jack type.

marked "DC —" is used for all d.c. ranges so is called the "common terminal"; one test lead is plugged here while the other is placed in the current jack marked for the desired range.

Resistors R_4 to R_7 are voltage multipliers. The current shunt resistors R_1 , R_2 , R_3 remain in the circuit to give a basic 1-ma. meter movement (sensitivity is 1000 ohms-per-volt). Again the "DC —" jack is the common terminal; one test lead goes here while the other is plugged into the desired d.c. voltage jack.

For a.c. voltage measurements, switch SW_1 is thrown to the AC position, thus inserting a copper-oxide reca correction resistor, used to adjust the a.c. range according to the meter and copper-oxide rectifier characteristics.

Although the ranges of the d.c. and a.c. voltages are the same, they do not usually use the same meter scale, because the rectifier characteristics change somewhat with the amount of current flowing through it. This usually means that the a.c. scale will be crowded near zero. Separate scales are therefore on the meter face for each purpose, direct current and voltage, a.c. voltage, and resistance. A typical scale is shown in Fig. **30**.

Both series and shunt ohmmeter ranges are provided. The "Common Ohms" jack is one terminal for all ohmmeter ranges. Switch SW_2 is closed for shunt ohmmeter measurements. (This switch must be open for all other measurements.) This places R_{13} and battery B_1 in the circuit. The "500 φ " jack is used for this shunt ohmmeter.

There are two series ohmmeter ranges, using a tapped battery B_2 and separate zero adjustments. The a.c.d.c. switch must be in the d.c. position for all ohmmeter measurements.

Practical multimeters of this type may have fewer or different ranges, but the circuits will be essentially like the one just described.

SELECTOR SWITCH TYPE

This multimeter eliminates most of the jacks by using a multiple selector switch to select the desired range and function. A typical one is shown in Fig. S1. Switches SW_1 and SW_2 are ganged together. In position 1, the highest voltage range is obtained, through multipliers R_1 , R_2 , R_3 and R_4 . Notice that only switch SW_2 is in the circuit at this setting, while at other settings SW_1 is in alone or in combination with SW_2 .

In positions 2, 3 and 4 of the selector switches we have the remaining voltage ranges; positions 5, 6 and 7 are the current ranges; position 8 is a shunt ohmmeter and position 9 is a series ohmmeter. Other ranges can be provided by adding other switch contacts.

Note that the same multiplying resistors are used both for a.c. and d.c. voltages. This is possible where similar ranges and sensitivities are provided. Of course, the a.c. ranges will be marked on different scales.

Where the d.c. voltage ranges are at sensitivities greater than 1000 ohmsper-volt, it is standard practice to shunt the a.c. ranges so they remain 1000 ohms-per-volt. This requires separate multipliers for d.c. and a.c. and usually requires more switch sections.

The test leads are plugged into the jacks marked "DC—" and "+" for d.c. voltage, d.c. current and ohmmeter measurements. The jacks "+" and "AC \pm " are used for a.c. voltages, while the jacks "+" and "output" are used for a.c. readings when d.c. must be blocked out. This jack is marked "output" because it is usually used for making output voltage measurements in the plate circuit of the power tube.

Tolerance. How accurate do our measuring devices have to be? The answer depends on the type of work being done.

In a research or design laboratory, highly accurate measurements may be desired. Here, expense usually does not matter; it is worth the price of handcalibration and extreme care to make sure of success in the particular work being done. Further, precision parts



FIG. 30. Typical multimeter scales. Don't worry about reading this one—you will get instructions with any meter you buy. From the top down, the scales are: series ohmmeter; shunt ohmmeter; d.c. voltage and current; and last, a.c. voltage.

are usually used in research, so the observed results can be checked closely by calculations.

If this research is being carried on for a radio set manufacturer, some particular circuit design may be developed. Then, parts will be varied over wide limits to see just what variations can be permitted and still obtain rea-



FIG. 31. A multimeter of the selector switch type.

sonably good results. There is a very important reason for this-the question of cost. Precision parts are practically hand-made, and therefore quite expensive. But if wider tolerances are permissable, it will be possible to use cheaper mass-production parts, thus lowering the cost of the set.

When a radio set is made from parts having tolerances of 10% to 20%, the manufacturer is interested in over-all results, not the exact happenings in each stage. Thus, it is possible for readings to be off 30% to 40% from the expected value in many instances, yet the receiver may give average results for its type. Therefore, there is little reason for the serviceman to use

meters having an accuracy better than 2%. Far wider variations will arise in the radio circuit itself.

One of the things you must learn in service work is to interpret the readings obtained, considering the circuits where measurements are being made. In one circuit, variations of 50 volts or so may be expected, while in another a few volts variation may indicate trouble. After gaining a little experience, your growing knowledge of circuits and their operation will soon help you determine just what to expect. Just remember that a meter cannot point out the trouble; it can only tell you what circuit conditions are, from which you in turn figure out the cause.

THE N. R. I. COURSE PREPARES YOU TO BECOME A RADIOTRICIAN TELETRICIAN R REGISTERED U.S. PATENT OFFICE (REGISTERED U.S. PATENT OFFICE)

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Lesson Questions

Be sure to number your Answer Sheet 28FR-3.

Place your Student Number on every Answer Sheet.

Send in your set of answers for this lesson immediately after you finish them, as instructed in the Study Schedule. This will give you the greatest possible benefit from our speedy personal grading service.

- 1. Is the resistance of a shunt used to extend a 1-ma. meter to 100 ma. more, less, or the same as the shunt used to extend the range to 10 ma.?
- 2. Will the current indicated by a milliammeter in the cathode circuit of a screen grid tube be an accurate measurement of the plate current?
- 3. Give the five rules for connecting a current meter. 1. Series Committee 2. Contract Paint 3. Current Range 4. Muter Restance 5. Polarity
- 4. What will happen if you attempt to measure 60-cycle a.c. with a D'Arsonval meter alone? you will get the autrage current of the autrage current of the autrage current of time (300)
- 5. Which types of meters can be used to measure high audio frequency currents accurately, regardless of wave form? HOT WIRE '74 RS
- 6. Suppose a copper-oxide rectifier voltmeter reads about 55 volts when you are checking a 110-volt power line. What is the probable trouble?
- In measuring voltage in high-resistance circuits, will a 20,000 ohms-per-volt meter give more accurate circuit voltage measurements?
- 8. Which type of ohmmeter, when turned on, draws current from its battery even though the test leads are not touching each other?
- 9. Is the "zero" adjustment for a shunt type ohmmeter made when the test leads are held together or when they are separated?
- 10. Suppose you are measuring in a circuit where both a.c. and d.c. exist together and you wish to measure only the a.c. Would you use: 1, the "AC \pm " multimeter jack; or 2, the "OUTPUT" jack, to exclude the d.c.?

World Radio History

DIPLOMACY

Diplomacy in its true sense is simply courtesy. A true diplomat is one who shows a deep appreciation of the other fellow's feelings.

A diplomat knows when to agree with you—and how to disagree. He will agree on trivial matters regardless of his own opinions, thereby winning your good will. When it comes to something important, however, he will bring you to his way of thinking *painlessly*. You will enjoy doing what he requests because he makes you feel that you are an important person.

Now turn matters around—and you be the diplomat. In conversation, be considerate of the other fcllow's feelings and pet beliefs. Don't contradict people flatly—nobody appointed you to go around correcting the mistakes of others.

So learn diplomacy—and practice it in all your contacts with people.

J. E. SMITH



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STUDY SCHEDULE NO. 29

For each study step, read the assigned pages first at your usual speed. Reread slowly one or more times. Finish with one quick reading to fix the important facts firmly in your mind, then answer the Lesson Questions for that step. Study each other step in this same way.

🗹 I. A Basic D.C. Vacuum Tube Voltmeter Pages 1-5

The vacuum tube voltmeter is a radio stage with a means of indicating the voltage fed to it. Considerable ingenuity is used in getting the indications accurately and to eliminate effects of the v.l.v.m. on the circuit where voltages are to be checked and vice versa. You study a simple d.c. type to learn some of the problems and how they can be overcome. Answer Lesson Questions 1, 2 and 3.

Improvements in sensitivity, so small voltages could be measured, led to the development of several more elaborate types for d.c. measurements. The bridge type you will study is one of the leading circuits in v.t.v.m. service instruments today, so study this section carefully. Answer Lesson Question 4.

We go back to our basic d.c. circuit and learn how it can be adapted to measure a.c. Many new problems now come up, particularly restrictions on the frequency range caused by test leads. Even the calibration requires care, as the wave shape will upset results. Answer Lesson Questions 5 and 6.

4. Peak Vacuum Tube VoltmetersPages 13-17

Again, sensitivity needs lead to improved circuits—where not only more sensitivity, but freedom from test lead effects can be obtained. The diode rectifier type and the slide-back circuits in particular are important. Answer Lesson Question 7.

Here we take up vacuum tube indicators which use several stages of amplification. The tuned r.f. types are the "signal tracers" used by many service shops because of their ability to trace the signal through a radio, thus leading right to the trouble. There is also an important section on calibration of the v.t.v.m. —a procedure necessary at frequent intervals to maintain accuracy. Answer Lesson Question 8.

- 6. The Cathode Ray Oscilloscope Pages 20-28 This is a truly remarkable instrument—one which lets you "see" the signal. Here you learn how we get a spot of light, how to move the spot and how to make an exact tracing of the signal. Answer Questions 9 and 10.
- 7. Mail Your Answers for this Lesson to N.R.I. for Grading.

8. Start Studying the Next Lesson.

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VACUUM TUBE VOLTMETERS

CATHODE RAY OSCILLOSCOPES

A Basic D. C. Vacuum Tube Voltmeter

V/HILE nearly all kinds of service work can be done with a standard service multimeter, some of the more difficult jobs can be done more quickly by using special instruments, such as the vacuum tube voltmeter and the cathode ray oscilloscope. For years, these instruments have been standard laboratory and factory equipment. As they are particularly suited for many special jobs, more and more of them are finding their way into the better equipped service shops. Especially as television and high-fidelity receivers become more widespread, there will be an increasing need for knowledge of the uses and limitations of these instruments. Both will be covered in this lesson.

► We will start with the vacuum tube voltmeter—originally developed as an r.f. voltmeter but which was soon used for d.c. measurements as well.

You recall that ordinary voltmeters affect the circuit whose voltage they measure—because they draw current and because they add the meter resistance to the circuit. Fortunately, in ordinary service work these effects are not usually serious, or can be allowed for. In other fields, however, they often make involved calculations necessary.

Even in a simple circuit like Fig. 1, the voltmeter will not indicate the true voltage across R unless its resistance is many times higher than that of R, as the voltmeter current through R_1 causes an additional voltage drop. As you have learned, the voltmeter ohmsper-volt sensitivity determines the extent of the difference between the voltmeter reading and the actual voltage when the meter is removed.

Nowadays, standard meters have sensitivities of 1000 ohms-per-volt, which is acceptable for ordinary service work. For measurements in circuits containing reasonably high resistances, higher sensitivities up to 25,000 ohms-per-volt are available. But where very high resistances are used, as they are in so many radio applications, increased voltmeter sensitivity is needed. That's why d.c. vacuum tube voltmeters are used.

BASIC TRIODE V.T.V.M.

Fundamentally, a vacuum tube voltmeter is just an ordinary radio tube stage having an indicator instead of the usual plate load. Figure 2A shows a typical triode circuit. An E_g - I_p characteristic curve for this type tube is shown in Fig. 2B. By using the proper grid bias, we can operate on the straight part of the characteristic





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(point 1), over a curved portion (point 2), at cut-off (3), or even beyond plate current cut-off (4), depending upon just what circuit action we want.

Suppose we operate on the straight part of the characteristic, the tube then acting as a class A amplifier. There will be a no-signal plate current, indicated by the meter in the plate circuit. We can make the meter read mid-scale for this current, either by using a shunt resistor across the meter or by adjusting the plate and grid voltages.

Now if we connect the test probes to a d.c. voltage source, current will flow through R_1 , causing a voltage drop across it. Since this voltage drop is between the grid and cathode, it will either increase or decrease the plate current, depending on the polarity of the drop. By calibrating the plate current meter properly, we can make it indicate the voltage of the source that R_1 is connected across, so we can make a simple d.c. vacuum tube voltmeter out of a standard radio tube amplifying circuit.

CIRCUIT LOADING

Why is our vacuum tube voltmeter (or "v.t.v.m.," as radio men call it) any better than an ordinary meter? We put the resistor R_1 across the source of voltage being measured—just as we'd have put the meter resistance across the source if we used an ordinary voltmeter. The difference is the fact that R_1 can be made very large. The only real limit on the size of R_1 is the grid current which may result from contact potential and gas within the tube. If resistor R_1 is too large, these tiny currents would cause a positive voltage drop across it which could nullify the C bias and cause erratic changes in plate current. However, resistor R_1 can be as high as 10 or 15 megohms. Just compare this with the

few thousand ohms introduced by the ordinary meter, and you can see that our v.t.v.m. will have far less effect on the circuit whose voltage we're measuring!

For example, if resistor R_1 is 15 megohms and 3 volts across it will give a full-scale meter deflection, the sensitivity of our v.t.v.m. is 5 megohms-per-volt. Commonly, sensitivities of from 1 to 5 megohms-per-volt are obtained on the basic range in commercial types. This is obviously many times higher than even the best 25,-000 ohms-per-volt D'Arsonval meter.

• Of course, the simple circuit shown in Fig. 2 needs improvement. The calibration depends upon the characteristics of the particular tube used; no provisions are shown for extending the range; the device will be affected by any stray a.c. voltages; and the circuit does not make full use of the meter sensitivity. Let's see how these limitations can be cured.

CALIBRATION LINEARITY

No two radio tubes (even of the same type) have E_{g} - I_{p} characteristics which are exactly the same. Since the amount of plate current change for a particular applied grid voltage depends upon the shape of this curve, our v.t.v.m. would probably have to be recalibrated each time the tube burned out and a new one was put in. Furthermore, the shape of the characteristic is greatly affected by the supply voltages. As the batteries run down, the calibration will be thrown off.

One way to solve this difficulty is to use a relatively high plate load resistance in series with the meter. This makes the tube characteristic more nearly straight and less dependent on small supply voltage variations.

An even better scheme is to put a self-bias resistor (like R_2 in Fig 3) in the cathode lead. Now, with this ar-

rangement, if the plate current drops because of tube aging or battery supply reduction, the bias voltage will also drop. This tends to keep the plate current more nearly constant. As this bias changes with plate current variations caused by the applied voltage, this produces a bias which tends to oppose the applied voltage. The tube is forced to act as if it had a lower amplification factor, but the stage characteristics are made relatively independent of the tube and supply voltages. This method (called degenera-



FIG. 2. An ordinary amplifier stage makes an excellent d.c. vacuum tube voltmeter, as the plate current is proportional to grid voltage.

tion) sacrifices some sensitivity, but the benefits outweigh the loss.

KEEPING OUT A.C.

Since the circuit in Fig. 2 is an amplifier, any a.c. voltage applied across resistor R_1 will cause an a.c. variation of the plate current. This will not affect the D'Arsonval meter, which indicates only the average d.c. plate current. However, if this a.c. exceeds the bias, the grid will swing positive and grid current will flow, which will cause false readings.

Figure 3 shows how a.c. troubles can be avoided. An extra resistor R_3 and a condenser C have been added in the grid circuit. These parts act as an



FIG. 3. The filter C-R₃ prevents a.c. ripples from affecting the d.c. voltage readings.

a.c. filter. The applied a.c. is dropped across R_3 , as its resistance is large compared to the low reactance of C. This reduces the level of any a.c. input voltages at the grid of the tube. Should these voltages still be high enough to swing the grid positive, R_1 - R_3 -C act as a grid leak and condenser, automatically biasing the tube. This scheme lets us measure the d.c. voltage in a circuit containing both a.c. and d.c., without the a.c. affecting the reading.

EXTENDING RANGES

Since we might want our v.t.v.m. to measure d.c. voltages ranging from just a fraction of a volt to 1000 volts or more, we need some means of extending the meter range.

Figure 4 shows the same method we used with an ordinary voltmeter. Resistor R_1 is the grid resistor of our vacuum tube voltmeter circuit and sets the basic range when the selector switch is in position 1. Then when the selector



FIG. 4. The addition of series resistances extend the range, like the multipliers used with ordinary voltmeters.

switch is moved to position \mathscr{D} or \mathscr{S} , the multiplying resistors R_2 or R_2 plus R_3 are added in series.

Since R_1 will probably be 1 or more megohms, R_2 and R_3 must have very high resistance values in order to extend the range. Very large resistances are not always available in compact, stable and exact sizes. Also, when the multiplying resistors are extremely high, any small amount of leakage across the selector switch or in the circuit will upset the voltage division.

► A better circuit is shown in Fig. δ' . When the selector switch is in position 1, resistors R_1 , R_2 and R_3 are all across the grid-cathode of the tube, and all the input voltage is fed into our v.t.v.m. Switch position 1, therefore, sets the meter to its basic range.

When the selector switch is moved to position 2, only part of the input voltage is applied to the input of the v.t.v.m., as the circuit acts as a voltage divider. The exact amount of division depends upon the ratio of resistors R_1 plus R_2 to the total of R_1 plus R_2 plus R_3 . Similarly, when the selector switch is moved to position 3, the voltage across R_1 will be all that causes the meter deflection. Proper choice of resistance permits any desired range extension.

This circuit has the advantage of using easily obtainable resistor sizes. Also, it presents a constant d.c. impedance to the supply source regardless of the range chosen (provided there is no excessive leakage in the grid circuit of the vacuum tube). The ohms-per-volt sensitivity of this v.t.v.m. goes down as the range is increased, but this does not greatly matter—it is still far higher than the average meter.

MEASURING CURRENT

Can a v.t.v.m. be used to measure ' current? Yes—very easily. Just insert

a known resistor in series with the circuit carrying the current and measure the voltage across the resistor with the v.t.v.m. Ohm's Law $(I = E \div R)$ does the rest. Of course, the known resistor should be made as small as possible so it won't have much effect on the circuit in which you insert it.

The range can be varied by using different resistors. For example, if 5 volts will give a full-scale reading, we can get a 50 ma. range by using a 100-... ohm resistor (.05 x 100 \pm 5 volts) or a 5-ma. range by using 1000 ohms.

MEASURING RESISTANCE

You can also measure resistance with a vacuum tube voltmeter. All you need is a known voltage source, as shown in Fig. 6. The resistor R_1 is a part of the v.t.v.m. and is, of course, a known resistance. The resistor R_x is the unknown one we want to measure. Here's how we do it.

First, the battery is connected between terminals A and B. This puts the battery right across resistor R_1 , and we get a deflection on our v.t.v.m. corresponding to the battery voltage. Usually the battery voltage is chosen or varied to give a full-scale meter deflection.

When the unknown resistance R_x is inserted in the circuit, the battery current flows through both R_x and resistor R_1 . This divides the voltage so



FIG. 5. A voltage divider method of extending the range.

the voltage drop across resistor R_1 is less than the total battery voltage. If resistor R_x exactly equals resistor R_1 , the voltage divides in half and we get a half-scale meter deflection. You remember that this is just how a regular ohmmeter works. Therefore, we can calibrate the v.t.v.m. scale in ohms. exactly as we do with any other ohmmeter.

The arrangement in Fig. 6 is much better than an ordinary ohmmeter for measuring high resistances. Resistor R_1 may be 10 megohms, which would give us a 10-megohm center-scale reading on our meter. Since we can read 100 times the center-scale value on an ohmmeter, this means we can readily measure unknown resistors as high as 1000 megohms, using only 3 volts or so as E. This means we can extend the range to far higher values and still use reasonable values of voltage. The method is the same we used with an ordinary ohmmeter-we add additional voltage and an additional resistance in series with it at point X. Thus, if the original circuit measures 1000 megohms with a 3-volt battery, a 30-volt battery will give a 10,000megohm range, while 300 volts will increase the range to 100,000 megohms. Compare this sensitivity with that of an ordinary ohmmeter, where 300 volts may give a range of only 10 or 15 megohms!

The limit to the amount we can extend our range is the leakage resistance across insulation between jacks, terminals and mountings for the various parts. To measure very high resistances, you must use extremely good insulating material and just as little of it as possible.

► Low resistances can also be measured with a v.t.v.m. by shunting R_1



FIG. 6. Using the v.t.v.m. as an ohmmeter.

with a small resistor or replacing it with a low ohmic value to lower the center-scale value. By setting the R_1 value at, say, 100 ohms, the mid-scale ohmmeter reading will be 100 ohms and we can calibrate our meter accordingly.

V.T.V.M. as a Multimeter. Since we can use a v.t.v.m. to measure voltage, current and resistance, it is possible to make a multimeter using this basic circuit. By using switches, the proper ranges and functions can be obtained.

This arrangement would give us an sensitive multimeter ---extremely which, however, would need frequent recalibrations and adjustments. Whenever the tube is changed, a complete recalibration may be necessary.

been letting the normal plate current

of the tube give our meter a mid-scale

Improved D. C. Vacuum Tube Voltmeters

The circuits so far covered have been simple. They will work satisfactorily in measuring fairly high voltages. Now let's see how better d.c. vacuum tube voltmeters can be made.

deflection. This has allowed us to neglect polarity in making measurements. However, it has also meant that we've Meter Sensitivity. So far, we've had to use a fairly insensitive meter.
For example, if the normal plate current is 10 ma., we must use a 20ma. meter to have the normal plate current cause a mid-scale deflection. Then, the applied voltage must cause a 10-ma. plate current change to give us a full-scale deflection. Even if the tube has a high mutual conductance, a 10-ma. plate current change requires a fairly high grid voltage, so the basic range of our v.t.v.m. is high. This means low voltages produce only small changes in plate current and are not easily read.

But if we can have our meter read zero when no voltage is being measured, we can use a much more sensitive meter—1 ma., for example. To do so, we must find some means of having zero plate current when there is zero grid input. How can this be done?

One way is to apply a higher bias and to work the tube at or near the plate current cut-off point. This is frequently done, but has the disadvantage of making the tube operate over a curved part of its characteristic. This means that plate current—and therefore meter readings—will not be linear for different applied voltages. Even worse, our meter will deflect least for small voltages, where we want the greatest sensitivity. A better plan is to let the tube operate on the straight part of its characteristic and use a bucking circuit to get zero plate current.

BUCKING CIRCUITS

Figure 7 shows one common bucking circuit. The normal plate electron flow of the tube (solid arrows) and the electron flow caused by the bucking battery E (dotted arrows) flow through the meter in opposite directions. Resistor R is carefully adjusted so that the current from E exactly equals the tube current. The meter then has equal and opposite currents flowing through it, so it reads zero. This is called a bucking circuit, as the extra current "bucks" or cancels the plate current through the meter.

Now when a voltage is applied to the grid of the tube, the plate cur-_ rent of the tube changes, while the bucking current remains the same. Then the meter indicates the difference between the two currents—in other words, the actual change in the tube plate current.

You can readily see that in using such a circuit we must be sure that the voltage applied to the grid makes it more *positive*. A more positive grid causes an increase in plate current, which will make the meter deflect upscale, as we want it to.

As you have learned, the important advantage of such a circuit is that we can use a more sensitive meter in the plate circuit, instead of the 10 or 20ma. meter needed in simpler vacuum tube voltmeters. If the tube has such a characteristic that a 1-volt change



FIG. 7. The bucking current cancels the plate current flow through the meter.

on the grid causes a 1-ma. change in plate current, and a 1-ma. meter is used, the basic range of our improved v.t.v.m. is 1 volt. (By using a more sensitive meter, we can get an even lower basic range.) Furthermore, this 1-volt scale is now spread over the entire meter scale instead of half of it, which alone doubles the scale sensitivity.

There are many variations of this bucking circuit, but all have disadvantages. If R is not adjusted almost perfectly, the meter may be overloaded and burned out. This means the meter must be shunted until R is adjusted. Further, the separate battery is a nuisance—among other things, its voltage decreases as it grows older. Several better circuits, using the bridge principle, have been developed. Let's see how they work.

D.C. BRIDGE V.T.V.M.

You will study the bridge circuit in more detail later, so here we'll just go into the basic principles briefly. A typical bridge is shown in Fig. 8.



FIG. 8. A resistance bridge.

When a voltage is applied across terminals A and B, current flows through R_1 and R_3 , as well as through R_2 and R_4 . If we make R_1 equal R_2 and R_3 equal R_4 , the resistance of R_1 plus R_3 will equal R_2 plus R_4 , so the current through path A-C-B is equal to that through path A-D-B. This makes the voltage drop across R_1 equal to that across R_2 , so the potential of point C with respect to A is the same as that of D with respect to A. Hence, points C and D are at the same potential (no voltage difference between them)-and no current flows through the meter. The bridge is said to be "balanced" when the meter reads zero.

When either pair of resistors is made unequal $(R_1 \text{ not equal to } R_2, \text{ or} R_3 \text{ not equal to } R_4)$, there will be a voltage difference between C and D, so current will flow through the meter. The bridge is now unbalanced.

To use this bridge principle in our v.t.v.m, we set up the circuit shown in Fig. 9. This is much the same as the circuit in Fig. 8, except that we've replaced variable resistor R_3 with a tube and bias resistor R_6 . Resistors R_1 and R_2 are equal in value.

With no input on the grid, we balance this circuit by adjusting R_4 until no current flows in the meter. As you just learned, this means that R_4 equals the plate-cathode resistance of the tube plus the resistance of R_6 .

Now if we apply an input to the grid through terminals X and Y, the resulting grid voltage changes the tube plate current. This current change means the plate-cathode tube resistance has changed—so the bridge is no longer balanced and current flows through the meter. Calibrating the meter in terms of voltage input to the grid gives us our v.t.v.m.

Regulating the Power Supply. It is usually preferable to use a power pack for our bridge type v.t.v.m., instead of a battery. But such a power supply must be very well regulated. Changes in the supply voltage will shift operation of the tube to a different part of its characteristic and, as you know, this destroys the calibration of the v.t.v.m.

One common method of regulation,





7 World Radio History using a voltage regulator tube VR and a series resistor R_7 , is shown in Fig. 9. This tube has the property of passing a much higher current when the voltage across it increases slightly. The voltage drop this current causes in resistor R_7 then drops the voltage across VR. The net effect is to keep the voltage across VR, which is the voltage supplied to the v.t.v.m., very nearly constant. Under normal conditions, a modern regulator tube can maintain the voltage across its terminals within 2 or 3 volts of its rated value.

An Improved Circuit. An even better bridge type v.t.v.m. circuit is shown in Fig. 10. Here another tube, exactly like the first one, is used in place of resistor R_4 of our original bridge circuit.



FIG. 10. A balanced tube bridge.

The use of two tubes tends to cancel the effects of any change in supply voltage. In other words, if the supply voltage increases, the plate current to both tubes increases correspondingly, and the bridge remains in balance, thus eliminating the need for a regulator \lor tube.

▶ Resistors R_6 and R_8 are used to make the tubes bias themselves. This \lor also helps keep the circuit in balance.

An unusual feature of the circuit in Fig. 10 is the use of R_7 to get greater stability and higher sensitivity. Both plate currents flow through this re-

sistor, producing a voltage having the polarity indicated, so more self-biasing is produced. However, this bias voltage is high enough to cut the tube currents off except for the fact that it is _ bucked out by a positive voltage from R_{10} . (Trace the grid circuits to ground, to the junction of R_9 and R_{10} , then through R_7 to each cathode to see this.) Now the resulting net bias is about that of each cathode resistor.

When a voltage is applied to the input of VT_1 , so that terminal X is made positive with respect to terminal Y, the plate current of VT_1 increases. This causes an increased flow of current through resistor R_7 , thus biasing the grid of VT_2 more negatively. (The positive R_{10} drop remains fixed, but the negative R_7 drop increases, so the net effect is more negative bias on VT_2 .) Therefore, the plate current of tube VT_2 goes down.

This means that the plate resistance of tube VT_1 goes down at the same time the plate resistance of VT_2 goes



FIG. 11. The RCA-Rider Volt-Ohmyst Jr. is a typical d.c. type v.t.v.m. using the balanced bridge circuit.



up. Therefore, this bridge becomes much more unbalanced for a given input than does the circuit in Fig. 9, and so is more sensitive. Also, the v.t.v.m. in Fig. 10 can be made more linearreading than the Fig. 9 v.t.v.m. by choosing tubes whose characteristics balance each other.

An exact balance of the bridge (zero meter adjustment) is obtained by adjusting R_{11} . Varying this resistor varies R_1 and R_2 , adding to one and subtracting from the other. This varies the tube plate voltages, so the tube resistances change also. If the ratio of R_1 to R_2 is made the same as the ratio of R_3 to R_4 , the currents in the two paths will adjust themselves so that the same voltage drops occur, even though the currents are unequal. This ratio is usually given as $R_1 = R_3$

ratio is usually given as $\frac{R_1}{R_2} = \frac{R_3}{R_4}$

Notice that it is not necessary that $R_1 = R_2$ or $R_3 = R_4$, just so the ratio of values is properly chosen.

A COMMERCIAL BRIDGE V.T.V.M.

A typical commercial bridge instrument is the RCA Junior Volt-Ohmyst shown in Fig. 11. Its circuit is basically the same as that of Fig. 10, with provisions to extend ranges and measure resistance.

The Junior Volt-Ohmyst may be used as a multi-range d.c. type v.t.v.m., as a low- and high-range ohmmeter, or as an a.c. voltmeter. Selector switches permit changeover to the desired range and function. The a.c. voltmeter incorporated in the instrument is a standard copper-oxide rectifier typenot an a.c. vacuum tube voltmeter, so this is a d.c. type v.t.v.m.

A. C. Vacuum Tube Voltmeters

You recall that most standard a.c. voltmeters have distributed inductance and capacity which ruin calibration when you try to use them on highfrequency a.c. Highly sensitive a.c. vacuum tube voltmeters have been developed which are relatively free from frequency errors and limitations, even at radio frequencies. Let's see how they work.

A TRIODE A.C. TYPE V.T.V.M.

You learned that the basic d.c. v.t.v.m. shown in Fig. 2A won't indicate on a.c. because the tube is worked on a straight part of its characteristic. Thus, an a.c. grid input produces an a.c. plate current varying around the normal plate current—and the *average* of this a.c. plate current (which is all the meter can read) is just the same as the normal plate current. (See Fig. 12.) Therefore, the meter reads only the normal plate current as long as we operate the tube on the straight part of its characteristic.

But by changing the bias so as to move the operation to the curved part of the characteristic, or to plate current cut-off, we get the detector action shown in Fig. 13. Now there is little or no plate current when no signal is applied to the grid. When an a.c. voltage is applied, rectified pulses are produced which have an average different from the no-signal plate current. This difference in average plate current can be read on a meter.

Sensitivity. With this change in operating point, the circuit in Fig. 2A will read both d.c. and a.c. voltages, provided the d.c. voltage has a polarity which makes the grid positive. However, the meter scale is not at all linear, because of the curve of the tube characteristic near plate current cut-off. This means very little meter deflection is obtained for small voltages, so the scale is crowded and difficult to read at the low end.

As the applied voltage is increased, the meter deflection becomes more linear, so this circuit is satisfactory for fairly large voltages. Farther on



FIG. 12. The d.c. average of the plate current does not change when an a.c. voltage is applied to a class A amplifier.

in this lesson we'll take up circuits for measuring low voltages. \lor

Frequency Range. With the proper tube and circuit arrangement, we can measure very high radio frequencies on our a.c. vacuum tube voltmeter. The limit to which we can go before the scale calibration becomes inaccurate is determined primarily by the input circuit of the v.t.v.m. and the circuit where measurements are being made. The input capacity of the tube is in parallel with the grid resistance. As frequency goes up, the reactance of this input capacity goes down. This reduces the impedance between the input terminals, so our v.t.v.m. will eventually start loading the circuit across which it is connected.

This loading is not so serious when the v.t.v.m. is connected across a tuned

circuit. Since the v.t.v.m. is primarily a capacitive load, it will detune the resonant circuit. If we retune the circuit with the v.t.v.m. in place, we can automatically cancel this effect and make our measurements. Of course, the circuit must be retuned to the orig inal frequency when the v.t.v.m. is removed.

Test Lead Effects. In measuring r.f. voltages, it is desirable to use a shielded test lead to connect our v.t.v.m. to the circuit being measured. Shielding prevents the leads from picking up hum or other stray voltages which would produce false readings. However, the capacity between the shielded lead and the shield is connected in parallel with the tube input capacity. This increases the load introduced into the measured circuit, so



FIG. 13. By moving to the curved portion of the characteristic, there will be a d.c. average change from the no-signal value, when an e.c. voltage is applied.

a shielded lead further limits the frequency range.

Test Lead Resonance. The test leads become resonant whenever their physical length equals a quarter wavelength at the frequency where measurements are being made. At a frequency of 75 megacycles (4 meters), leads about 3 feet long would be a 1/4 wavelength line.

As you'll learn in another lesson, a quarter-wave line has an unusual voltage distribution. If we apply a high voltage to one end, we get very little or no voltage from the other cnd. Obviously, this would make our v.t.v.m. reading highly inaccurate!

While this quarter-wave effect occurs only at resonance, the test leads begin to affect the voltage at frequencies considerably lower. In general, a v.t.v.m. using test leads cannot be used



FIG. 14. The input circuit of an a.c. type v.t.v.m.

for voltages with frequencies greater than one or two megacycles, unless special calibration charts are provided. This is still quite an improvement over the standard a.c. voltmeter. Later on, we will see how voltages can be measured at higher frequencies.

Extending A.C. Ranges. The input circuit for the basic range of our v.t.v.m. is similar to the circuit shown in Fig. 14 where we have input resistance R_1 shunted by C_T and C_G , the test lead capacity and the tube input capacity respectively.

If we try to use the voltage-dividing circuit shown in Fig. 15 which worked so well for d.c., we find that C_0 causes trouble. Moving the selector switch to



FIG. 15. The tube capacity upsets the voltage division as the switch is rotated.



FIG. 16. This capacity voltage divider gives uniform a.c. voltage division.

position 2 shunts C_0 across resistors R_2 and R_3 . Since the reactance of C_0 varies with frequency, the impedance of C_0 in parallel with R_2 and R_3 will vary with frequency, while the resistance of R_1 remains constant. This means that the voltage division across the divider also varies with frequency—so the range we get at position 2 (or at position 3) depends on the frequency of the input voltage.

We can clear up this difficulty by using a capacitive voltage divider like that shown in Fig. 16. When used to measure a.c. voltages, the division of voltages across this divider is determined by the reactances of the condensers. In position 2, the a.c. voltage division is determined by the reactances of C_1 and C_6 . In position 3, C_2 is added to C_1 , so the combined reactance to that of C_{α} is the divider. Since these reactances all vary in the same manner when the input frequency varies, the division of voltages remains the same, and the various ranges of our v.t.v.m. are therefore very nearly independent of frequency.

This v.t.v.m. circuit may also be used to measure d.c. voltages. In this use, the condensers have no effect and the resistors determine the voltage division for the instrument's ranges, just as they did in the circuit shown in Fig. δ .

A.C. WAVE SHAPES

You recall that an a.c. cycle has a peak value, an average value and an effective value. Because we are usually most interested in the r.m.s value of a sine wave, we calibrate an a.c. type of v.t.v.m. to indicate r.m.s values.

But the meter of an a.c. type v.t.v.m. actually operates from the *average* of the plate current. Now the average plate current is not the same as the effective or r.m.s. value—so we have a meter which really operates on one value of an a.c. wave but is calibrated to indicate another.

This is perfectly all right as long as the ratio of the r.m.s. value to the average value is constant—as it is for any regular wave shape. If you want to determine the average value of the a.c., all you have to do is multiply the meter scale reading by the proper multiplying factor. The peak value can also be determined in the same way (using a different factor, of course).

But you must remember that the exact ratios between the r.m.s., average and peak values of a wave depend on the wave shape. Figure 17 shows the relationships existing in various kinds of waves. Notice that all these waves have a different peak-to-average-to- $\sqrt{}$ r.m.s. relationship. (The special wave of Fig. 17E is similar to television control pulses and has widely different r.m.s. and average values, depending on the height and width of the pulses as well as their spacing.)

Distorted Waves. The meter of a v.t.v.m. intended for service work is usually calibrated to read r.m.s. values when the instrument is measuring a pure sine wave. The readings may be in error if such a v.t.v.m. is used to measure a distorted wave, because the distorted wave will probably not have the same average-to-r.m.s relationship as a sine wave.

Suppose we start with a sine wave, but one receiver stage distorts the wave as shown in Fig. 18. One-half the wave is practically a square wave, while the other half remains a sine wave. This is commonly the result of an amplifying tube operating too near the current cut-off point.



FIG. 17. The shape and size of the wave pulse determines the relationship between the peak, average and r.m.s. values.

If we apply our test leads in such a manner that the sine portion of this wave makes the grid of the v.t.v.m. more positive, we'll get an indication which corresponds to the effective value of the sine wave.

On the other hand, if we reverse the test leads, the rectifying action of the v.t.v.m. will cut off the sine wave portion and we will get an indication corresponding to the distorted half of the cycle. This will give us an entirely different reading from the first one, because the relationship between the peak, average and effective values will be different. Distorted waves of this kind cause more error with a peak v.t.v.m. (to be studied later) than with average types (such as we have already discussed), because the peak is always very different from the r.m.s. value while the average and r.m.s. values are usually fairly close to each other.

▶ Reversing the v.t.v.m. leads may produce different readings even on an undistorted wave, because of the difference in capacity to ground between leads or parts in the v.t.v.m. The different readings caused by reversing leads is an effect known as "turnover."

Usually a shielded cable is provided to connect the v.t.v.m. to the radio being tested, with the shield used as one of the probes. Since the shield is always connected to an r.f. grounded point in the radio, the hot probe will always be the probe for measuring voltage, and it will not be possible to reverse the probes. Remember, however, that each amplifying stage reverses the phase of a signal 180°—



FIG. 18. A distorted wave, such as might be produced by operating a tube at the wrong point on the characteristic.

which produces exactly the same effect as reversing the leads—so it is still possible for distortion to produce an error in the reading as we go from stage to stage. Fortunately, in servicing, we are generally interested more in the *presence* of voltage than in the exact amount, so this problem is mostly limited to laboratory work.

Peak Vacuum Tube Voltmeters

A diode rectifier can be used like a copper-oxide rectifier, with a d.c. meter to make an a.c. voltmeter. The diode will give the voltmeter a wider frequency range, but otherwise has no advantages. Yet, a simple change in this diode circuit will make it one of the most valuable v.t.v.m. types.

PEAK-RESPONDING DIODE V.T.V.M.

The diode circuit just mentioned would have a bypass condenser across the meter, to prevent a.c. flow through the movement. By moving this condenser to the position shown in Fig. 19 and using a resistor R_1 , to set up an R-C time-constant circuit, we will have a peak-responding v.t.v.m.

When the plate of diode VT is made positive by the voltage source, condenser C is charged.

When the cycle reverses, the tube stops conducting and C starts to discharge through R_1 . However, R_1 is so large that condenser C does not have time to discharge very much before the cycle reverses and charges it up again. When fully charged, C has a voltage equal to the source voltage peak. After some of this charge has leaked off during the half cycle the tube does not conduct, the voltage drops somewhat. When the cycle reverses again, the tube can't conduct until the source voltage becomes higher than the reduced condenser voltage. When the tube does conduct, the condenser is charged to full source voltage almost at once.

The net result of this circuit action

is that C always has a voltage very near the peak voltage of the source. Since C is in parallel with the series combination of R_1 and the meter, this combination also has practically all the peak voltage across it all the time. This peak voltage causes a current flow through the meter-so we have a v.t.v.m. which indicates peak voltages. \blacktriangleright Unfortunately, R_1 has to be so high in resistance that very little current flows through the meter. A very small current change occurs for even large voltage changes so the meter does not make a good indicator. However, we can easily solve this problem.

If we remove the meter altogether, R_1 will have practically all the peak



FIG. 19. A simple diode peak-indicating a.c. type v.t.v.m.

source voltage across it all the time. You'll notice that the action of the tube and condenser makes the voltage across R_1 a d.c. voltage. Therefore, all we have to do is put a d.c. type v.t.v.m. across R_1 , and we'll have a sensitive way of measuring the peak voltage of the a.c. source.

Since this circuit draws current from the source only at the peaks of the applied voltage, this v.t.v.m. acts as if it had a very high resistance. When it is used to measure high frequency, the amount it loads the voltage source depends mostly on the capacity between the input terminals. The plate-cathode capacity of the tube is in series with condenser C, across the terminals, but condenser C offers low reactance to r.f., so the loading

depends primarily upon the internal tube capacity.

An Improved Circuit. An even better circuit is shown in Fig. 20. Here's how it works.

During the half of an input a.c. cycle that makes terminal A positive with respect to terminal B, the diode conducts. Electrons flowing through the tube charge C_1 to the peak source voltage, with the polarity shown.

When the input cycle reverses and the tube ceases to conduct, the voltage on the condenser and the input voltage combine to cause an electron flow through R_1 . However, the time constant of C_1 and R_1 in series is so high that C_1 discharges only a little during this time. C_1 is thus only a little below the source voltage when the cycle reverses again. As soon as the source voltage rises above the voltage on C_1 , the tube conducts again, and C_1 is rapidly charged up to full peak voltage.

Thus this circuit acts in much the





same way as the previous one—almost the full peak voltage remains on C_1 all the time, and the circuit draws current from the source only when the peak value exceeds the C_1 voltage. Thus, the circuit draws but little current from the source.

▶ The voltage indications are obtained from the d.c. type vacuum tube woltmeter which is connected through filter R_2 - C_2 so that it measures the voltage across R_1 . Any r.f. across R_1 is prevented from getting to the v.t.v.m. by the a.c. filter R_2 - C_2 . On the half cycle when terminal Bis positive, the voltage across C_1 and the source voltage combine to force a small current through the high resistance, R_1 . The resulting voltage drop across R_1 has a d.c. average which operates the v.t.v.m. On the other half cycle, the source voltage and that across C_1 buck or oppose each other. Hence, there is practically no current through R_1 on this half cycle.



FIG. 21. An acorn type tube, mounted right in the probe, provides a minimum of input

in the probe, provides a minimum of input capacity, allowing measurements up to several hundred mc.

At low frequencies, the time between reversals of the input voltage may become so great that nearly all the charge leaks off C_1 . The circuit then ceases to be a peak-indicating device. However, it is easy to give C_1 and R_1 such a high time constant that the frequency at which this effect occurs is quite low.

▶ The amount this circuit loads the measured circuit depends on the platecathode capacity of the tube and on stray capacity between the terminals and between C_1 and ground. By using a tube with extremely low capacity, such as one of the modern acorn type tubes, we can reduce this capacity so that very high frequencies can be measured. In fact, we can mount the tube right in the test probe, and so eliminate most of the difficulties with test leads that you learned about earlier in this lesson. Figure 21 shows a tube mounted in this fashion. Using such a probe and reducing stray capacities in the circuit to a minimum, we can measure voltages at several hundred megacycles with this instrument.

There are three important things to remember about this type of v.t.v.m.:

1. It can be made fairly sensitive.

2. It can be used at extremely high frequencies.

3. It will not load the measured circuit much.

► It is possible to change the range of a v.t.v.m. of this type by using a multi-range d.c. vacuum tube volt- V meter.

► The peak-responding type of v.t.v.m. is more sensitive than average-



Courtesy General Radio Co.

A vacuum tube voltmeter capable of measuring r.f. voltages up to several hundred megacycles. An acorn type tube is mounted right in the probe unit to reduce input capacity and make such high-frequency measurements possible.

indicating types, because the peak value of a wave is almost always reasonably high and so provides enough voltage to operate the meter—while the average voltage may be quite low. However, since a peak v.t.v.m. is usually equipped with a meter calibrated to indicate r.m.s. or effective values, it may give incorrect readings on distorted waves.

SLIDE-BACK V.T.V.M. CIRCUITS

There is another kind of v.t.v.m. used to measure peak values. It is known as the "slide-back" type.

A typical circuit is shown in Fig. 22. In using it, the slider of potentiometer P is first moved to point 1. The voltmeter $V_{\rm M}$ then reads zero. Next, with no signal applied (terminals A and B shorted together), the small tube current is read on meter $I_{\rm M}$. This is not quite zero, because contact potential within the tube causes a small amount of current flow. The exact *amount* of current does not matter, just so we remember it.

Then the voltage to be measured is applied between terminals A and B. This causes a current flow through the meter $I_{\rm M}$ and through resistor R_1 . Resistor R_1 and condenser C_1 can be so chosen that this current will cause a meter deflection proportional to the peak value of the applied voltage.



FIG. 22. A positive-peak slide-back v.t.v.m.

Now, with the signal still applied, potentiometer P is adjusted to bring the pointer of current meter $I_{\rm M}$ back to its initial position. In other words, the potentiometer P is used to introduce a bucking voltage from the battery into the circuit. When this bucking voltage just brings $I_{\rm M}$ back to its initial reading, it is equal to the peak signal voltage. Then all we have to do is read the bucking voltage on meter $V_{\rm M}$, and we know what the peak voltage is.

This circuit has several advantages. We don't have to calibrate meter I_{M} , because we use only its initial pointer position. Also, the characteristics of the tube and other parts aren't very important. In fact, the accuracy of the measurement in a properly adjusted circuit depends only on the accuracy of the d.c. meter V_{M} .

Some precautions must be taken in using this slide-back meter, however. Principally, you must be sure to adjust the potentiometer just enough to bring $I_{\rm M}$ back to its initial reading and



FIG. 23. A negative-peak slide-back v.t.v.m.

no farther, to prevent errors in the reading.

Negative Peak Indicator. The circuit in Fig. 22 works whenever terminal A is made positive with respect to terminal B. Therefore, this device indicates on d.c. as well as a.c.

Should the polarity be reversed or should we want to measure the other half of the a.c. cycle, we can reverse the terminals A and B. Sometimes it is not a good plan to do this, as there may be capacity effects that would affect the readings. Also, it may be necessary or desirable to ground terminal B.

These possible difficulties make it better to leave terminals A and B connected as before, and change the circuit as shown in Fig. 23. Here the diode, the meter I_{M} and the slide-back battery have been reversed. Now, when terminal A is negative (B is positive), the diode passes current and the instrument works as before—except/ that it measures the negative peaks.

Using Slide-Back Voltmeters. As

there must be a small tube current so that the meter $I_{\rm M}$ will indicate, there is some rectification of the extreme tips of the a.c. cycle. This produces an error of about .5 volt, which is appreciable on low-voltage measurements but is negligible on high voltages. This error must be allowed for when measuring voltages below about 25 volts.

► To protect the instrument, always operate it in the following manner:

Short the input terminals and make a note of the initial $I_{\rm M}$ reading. (All practical types have an extra biasing battery and potentiometer with which you can adjust the initial current to a small value.) Then apply the maximum slide-back battery voltage before connecting to the unknown voltage source. After the unknown voltage is applied, adjust the potentiometer until meter $I_{\rm M}$ rises from zero to its initial reading. This procedure protects the current meter.

Triode Slide-Back V.T.V.M. You can use a triode tube as a slide-back



v.t.v.m., as shown in Fig. 24. With terminals A and B shorted, the C bias is adjusted to give some particular plate current I_{M} . Then the maximum slide-back voltage is introduced by potentiometer P. With the unknown voltage now applied to terminals Aand B, the potentiometer setting is reduced until the meter I_{M} returns to its original reading. The V_{M} reading is then equal to the peak voltage applied.

This circuit can only be used to measure positive peaks. However, it causes less loading of the source of voltage, as the grid circuit draws no current from the source.

Special Vacuum Tube Voltmeters

Some of the peak vacuum tube voltmeters we've studied can be made to have a basic range of about 1 volt,

TUNED RADIO FREQUENCY V.T.V.M.

which is sufficient for many r.f. measurements. However, to measure the small voltages in the r.f. stages of a receiver, we need a much more sensitive instrument.

Furthermore, all the a.c. types studied so far have one fault. They measure any input signal, whether it's the one we want or not. Noise pulses, hum, or additional r.f. voltages that happen to be in the circuit will be measured right along with the signal whose voltage we're trying to measure. Now, we can overcome this difficulty by using a tuned circuit. As you know, a resonant circuit has the ability to separate frequencies. By tuning a resonant circuit to the desired frequency and using it in our v.t.v.m. ahead of the d.c. indicator, we can make measurements at just the frequency we want.

A resonant circuit also makes it possible to use several amplifying stages ahead of the rectifier in our peak v.t.v.m., coupling from stage to stage just as in a radio receiver. This, of course, gives us the extra sensitivity we need for measuring small voltages.

► What kind of amplifier shall we use? A superheterodyne? No-because

the superheterodyne will itself produce interfering signals unless extreme care is exercised. Instead, we use tuned radio frequency stages in our amplifier. Although the stage gain with a t.r.f. circuit is not as great as with a superhet, the former gives us a more reliable instrument.

▶ We don't use a tuned circuit at the input of the vacuum tube amplifier. If we did, as soon as we connected the input across the radio part whose voltage we wanted to measure, the tuned circuit would either be detuned by the reactive component of that part, or be so loaded by the resistive component that the Q of the circuit would drop. Either condition would cause a loss of amplification, and so make the calibration of our v.t.v.m. inaccurate.

Instead, we use a special untuned input circuit which feeds into an amplifying tube. The tube then feeds into a tuned coupler which connects to the next tube in line. The special input circuit, by sacrificing some amplification, eliminates the effects of the shielded cable and tube input capacities. This means the input loading is reduced and the input will not detune the signal source.

Figure 25 shows the input circuit. Capacity C_{T} represents the input tube capacity combined with the shielded cable capacity. We'll first analyze its operation with the switch SW in position 1.

The circuit is connected to the point of measurement by two leads. One is a grounded lead (usually clipped to the chassis of the radio under test). The other is a shielded cable about 3 feet long with a test probe at the end. The test probe of the shielded cable is the secret of this device. Inside the insulating sleeve of the probe is a small mica condenser C_1 , usually between 1 and 4 mmfd., which connects the probe point P and the "hot" inner cable lead.

This probe capacity is in series with the signal source and capacity $C_{\rm T}$. Since C_1 is much smaller than $C_{\rm T}$, and capacities in series give a value less than the smallest, the maximum capacity across the circuit where measurements are made will be approximately the probe capacity. This value is so small that practically no detuning of the signal source will occur at low or medium radio frequencies.

When probe point P is connected to an r.f. or i.f. terminal, the source voltage divides between the probe capacity C_1 and capacity C_T . If the input capacity C_T is 100 mmfd. and the probe-to-cable capacity C_1 is 2 mmfd., 1/50 of the source voltage is applied to the tube input. If the amplifier has a gain of 50,000, the net over-all gain of the v.t.v.m. is 50,000 \div 50, or 1000 times. Thus, we sacrifice some of the gain to use the probe capacity and avoid detuning and loading effects but we still have considerable amplification left.

The voltage division will remain the same for all frequencies which are low enough to allow the cable to act essentially as a capacity (no resonant effects). The average limit is about 5 mc. Since most of the service measurements normally made with this instrument will be below 2 mc., this is not much of a limitation. In a factory-built unit, the tuning range is restricted to those frequencies where reasonably flat response can be obtained.

The probe capacity allows us to use a capacity voltage divider to obtain extra ranges. Turning switch SW to position 2 puts C_2 in parallel with C_T . If C_2 is 900 mmfd. and C_T is 100 mmfd., the total shunting capacity is now 1000 mmfd. Assuming the probeto-cable capacity C_1 is 2 mmfd., the voltage division is now in the ratio of 2 to 1000, so only 1/500 of the actual voltage is applied to the input tube. This means we've increased the voltage range to 10 times what it was in position 1.

The output of this first stage feeds into a resonant circuit, which in turn feeds the second tube. It is possible to cover a wide frequency range by using plug-in coils or a wave band switch in the resonant circuits.

► This circuit still offers some problems. First, the amplification obtained over a wide frequency band is not constant, but varies according to circuit conditions. Further, the amplification obtained from a tuned circuit varies with the age of the parts, because moisture absorption and physical changes upset the inductance, capacity



FIG. 25. The input circuit of a tuned circuit v.t.v.m. This is the basic circuit used in signal tracers.

and Q factor values. Finally, the amplification of a tube itself varies with age and with supply voltage changes. \checkmark

You may wonder just what amplification has to do with the v.t.v.m. reading. The answer is—our v.t.v.m. is calibrated with the amplifier in place, and a change in amplification changes the calibration. For example, suppose our amplifier gives us an amplification of 1000. We apply a certain signal voltage and get, say, a fullscale deflection. Then the amplification changes to 800 for some reason. Now if we apply the same signal voltage, we get—not a full-scale deflection -but only .8 of full scale. If our scale is calibrated on the basis of a 1000 amplification, obviously our reading is going to be wrong when the amplification changes to 800.

Since extreme accuracy is not needed in service work, however, many instruments of this type, known as signal tracers, have been developed and manufactured. They indicate voltage with reasonable accuracy and are very good for comparative purposes.

For example, if the voltage at the input of a radio stage is measured and found to be 2 volts and then the voltage at the output is found to be 10 volts, the gain of this particular stage would be $10 \div 2$ or 5. If there is any error in the reading of the voltage by the signal tracer, the same error will be in both these readings, and the errors will cancel. Thus, the actual gain of a radio stage can be determined with a tuned v.t.y.m., even it neither of its voltage measurements is accurate. Incidentally, for greatest accuracy in making comparative measurements of this sort, always make them with the switch SW set to the same range for both measurements if possible.

CALIBRATING VACUUM TUBE VOLTMETERS

A vacuum tube voltmeter must be recalibrated at frequent intervals, because of tube and voltage changes. Commercial instruments are provided with adjustments for this purpose.

In general, you calibrate a v.t.v.m. by applying a known voltage input, then adjust the instrument until it reads correctly. For d.c. vacuum tube voltmeters, ordinary dry cells are usually used as the known voltage source. The battery voltage should be measured with an accurate d.c. voltmeter while calibrating the v.t.v.m., however, as brand-new batteries may be as much as 10% higher than their rated voltage. Naturally, any error in the calibrating voltage will cause a similar error in the v.t.v.m.

When an a.c. vacuum tube voltmeter can be calibrated on 60 cycles, the circuits shown in Fig. 26 can be used. The accuracy depends on the standard meter—if $V_{\mathbf{M}}$ is a copperoxide type the accuracy may only be about 5% unless the meter has been hand-calibrated or checked against a standard.

Figure 26A shows a simple circuit which can be used if the test voltage is near the full-scale reading of $V_{\rm M}$. (As you will recall, meters are more accurate at full scale.) If the test voltage applied to the v.t.v.m. must be considerably lower than full scale of the meter, you should use the circuit of Fig. 26B. Here, rheostat R is adjusted to give $V_{\rm M}$ a nearly full-scale deflection. Then the desired test voltage is taken from voltage divider R_1 - R_2 .

The accuracy with which the test voltage is known in Fig. 26B depends on the tolerances of R_1 and R_2 , as well as on the meter V_M . Precision resistors, within $\frac{1}{2}$ % to 1% of rated values should be used.

If the meter cannot be calibrated on 60 cycles, a frequency within the range of the device must be used. A signal generator, with an amplified output, might be used as the source. The meter will have to be a thermocouple type, since that is the only



FIG. 26. Two methods of getting an a.c. voltage for calibration purposes.

standard meter having any accuracy for r.f. measurements. A circuit like Fig. 26B would be used, as the thermocouple cannot be used on very small voltage values. The resistors will have to be non-inductive (have no frequency errors) and the loading of the v.t.v.m. must be small.

Where a commercial instrument is being calibrated, the manufacturer's instructions should be carefully followed. They will usually tell you just how calibrations can be made with simple apparatus.

► Commercial instruments usually have an accuracy of from 1% to 5%. Should it be impossible to reach calibration, try a new tube in the v.t.v.m. You may have to try several tubes before finding one which will permit calibration.

The Cathode Ray Oscilloscope

Very often in radio work it would be extremely helpful if you could actually see the form of the a.c. waves you're working with. You seldom have a pure sine wave of current or voltage, you know—the radio circuits always distort them somewhat (and sometimes a great deal). Figures 17 and 18 show some of the wave forms you might meet in a radio or television set.

Now, a.c. meters are almost always calibrated on the assumption that they are going to measure pure sine waves. Va cathode ray oscilloscope does.

If they're used to measure distorted waves, their readings will usually be incorrect. Often this doesn't make much difference-but sometimes it's very important to know just what kind of a wave form you're measuring.

If you go into radio design, you'll find it necessary to draw graphs of resonance curves, frequency response, phase shift, and many other circuit characteristics. Such curves can be drawn by making many measurements and doing a lot of laborious workbut it would certainly be a lot easier if there were an instrument that drew them for you automatically!

There is such an instrument. It will let you see the exact shape of the wave form in radio circuits. It will draw the curves mentioned above for you-and even show you changes in them as they occur.

This remarkable instrument is called the "cathode ray oscilloscope." Every radio manufacturer-every radio laboratory - considers it indispensable equipment. More and more servicemen are finding it a valuable addition to their service bench.

In the rest of this lesson, you'll learn how the cathode ray oscilloscope works.

FLUORESCENT SCREEN

Certain chemicals, when bombarded by electrons, will give off a glow of visible light. This phenomenon is known as "fluorescence." It is the prin-

Now if we coat a transparent screen with these chemicals, and allow a thin beam of electrons to hit the screen, we will produce a dot of light on the screen. Further, if we can make the electron beam move over the screen in response to applied voltages, we can

make this dot of light trace out the wave form of the voltages. That's what

THE ELECTRON GUN

As you know, we can easily get a good supply of electrons by heating a cathode. If we put a positive plate with a hole in it near the cathode, the electrons will move toward the plate Some will hit the plate and others will go through the hole.

We can use this fact to produce the thin beam of electrons-or "electronic



Courtesy R. C. A

A typical c.r.o. designed for radio servicing. communications and dasign laboratory work.

pencil"-that we want to move over our fluorescent screen. Figure 27 shows how.

In Fig. 27A, a filament F with an oxide-coated tip emits electrons from a concentrated spot. (A special heated cathode is sometimes used in place of ciple on which fluorescent lamps work. V this filament.) These electrons are attracted toward a highly positive cylindrical plate P_1 . The filament is surrounded by another cylindrical electrode G. This electrode G is negative, and serves to concentrate the electrons and force them towards P_1 . It acts like the grid in ordinary tubes.

Since P_1 is highly positive, the electrons travel toward it at great speed and tend to go right down the length of the cylinder. Inside P_1 is a disc with a hole in it. Some electrons strike the disc and are collected by the plate, but most of them are concentrated in the middle of the cylinder and pass through the hole in the disc.

We now have a thin beam of electrons coming through the hole in the disc—the electronic pencil we've been looking for. But as this beam goes away from the disc, it tends to spread out. We need something to keep it together and focus it on the fluorescent screen.

The optical equivalent of our electronic pencil is shown in Fig. 27B. The lamp L acts as a source of light. This light is collected by reflector R and is concentrated in a single direction. The light rays would spread out as shown by dotted lines X and Y if it were not for the lens placed in their path. The lens serves to focus the rays to a point on the screen S.

▶ Returning to Fig. 27A, we find that cylindrical plates P_1 and P_2 act as an "electronic lens" to focus our electron beam. Plate P_2 is even more positive than P_1 , and is carefully placed with respect to P_1 . The electrons coming through the disc hole are speeded up by the highly positive potential of P_2 . Furthermore, an electric field exists between P_2 and P_1 .* This electric field collects and focuses the electron beam on the screen.

By choosing the proper P_1 and P_2 voltages, the electron beam is focused to a spot on the screen S. By varying the voltage on G, the number of electrons in the beam can be varied. In

*An electric field exists between any two objects having a voltage difference. This field is made up of electrostatic lines, which may be thought of as being similar to magnetic lines of force. Actually, magnetic fields are frequently used in cathode ray tubes for focusing. this way we can adjust the intensity of the light spot. This system is often called an "electron gun," because it "shoots" electrons at the screen.

The color of the light spot depends on the chemicals used in the screen. White, yellow, green or blue light spots may be produced by using different chemicals. At present, green is the most popular color for direct observation, and blue for photographing the image.

OBTAINING A DEFLECTION

Now that we have a spot of light produced by a pencil of electrons, we want to move the spot about on the screen and thus trace out wave forms.

This pencil of electrons consists of moving negative charges. If we place other electrical charges near this stream of electrons, we can divert its



FIG. 27. Comparing an electronic focusing method with a light system.

direction of flow, so we place a flat plate on each side of the electron beam between plate P_2 and the screen, as shown in Fig. 28. We don't want a difference in potential between these plates and P_2 , as this would make the electron beam spread out, so we connect plate B to P_2 .

As long as the flat plates are at the same potential with respect to each other, the electrons will travel onward in their normal direction, toward point 1 on screen S.

However, suppose we make plate A positive with respect to plate B. The

electrons in the beam will then be attracted toward plate A, and the beam will be deflected. The electrons do not go directly to plate A—they are traveling too fast. Instead, their direction is changed so that they strike the screen at point 2 rather than at point 1.

The amount the electron beam is bent from its normal direction depends upon the voltage applied between plates A and B. Of course, if the polar-



FIG. 28. How the electron beam is deflected by charged plates.

ity is reversed so that plate A is made negative with respect to B, the electrons will move in the other direction, toward point 3 on the screen.

Thus, placing these flat plates on each side of the electron beam gives us a means of making the electron stream move from point to point on the screen. The movement occurs as soon as the voltage is applied, so it will follow exactly any changes in voltage which may be impressed on the deflecting plates.

► Two sets of these plates are used one set to move the beam up and down and the other set to move it from left to right, as shown in Fig. 29.

Imagine that the electrons are coming out of the paper toward you. If there is no voltage on the deflecting plates, the electrons will make a spot in the center of the screen, as shown in Fig. 29A.

A voltage difference between plates c and d will cause a vertical (up and down) deflection, so these plates are called the *vertical deflecting* plates. If the voltage changes rapidly enough,

a vertical line will be formed, as shown in Fig. 29B.

Plates a and b cause a horizontal deflection, so are called the *horizontal deflecting plates*. An a.c. voltage between these plates will form a horizontal line, as in Fig. 29C.

Any a.c. above 20 cycles per second will produce these lines of light, because of the persistence of human vision. The eye continues to see light for a fraction of a second after it has been cut off, so the rapidly moving spot "blends" into a line.

Now that we can move the beam about on the screen, how do we get a complete picture of the wave form?

SWEEP VOLTAGE

Suppose we apply a sine wave voltage to plates c-d, with no voltage on plates a-b. The spot of light starts at



FIG. 29. By applying an a.c. voltage to one set of plates at a time, we change from a spot of light to either, a vertical or horizontal line of light.

the no-voltage position. (Fig. 29A), goes up until the voltage reaches a peak, goes down until the voltage reaches a negative peak, and repeats this motion as long as the voltage is applied. This gives us the vertical line shown in Fig. 29B, but it doesn't show us what the wave form looks like.

What we want is some means of making the spot move at a regular speed sideways at the same time it is moving up and down. Then the beam can trace out the wave form. Now we know how to make the spot move sideways—we apply a voltage to plates *a-b*.

You might think of applying the sine voltage to both sets of plates at the same time. However, this would give us a tilted line or a series of circular patterns instead of a picture of the wave we are interested in.

We want a voltage which will move the spot sideways until one cycle has been traced out, then sweep the spot back to the starting point to begin the process all over again. In other words, we want a voltage that increases regularly, then drops to zero when the correct point is reached.

Figure 30 shows just such a voltage. From a to b it increases regularly, then snaps down to c at once, and repeats the cycle from c to d to e. Because of its shape, this voltage is called a "sawtooth" wave. We call such a voltage a "sweep voltage" when we apply it to a



FIG. 30. The sweep voltage must increase, steadily and linearly to a maximum, then drop back quickly to the starting level.

cathode ray oscilloscope, because it sweeps the spot back and forth.

Sweep Generator. We can get this special wave from a circuit like that in Fig. 31. You recall that a condenserresistor combination has a time constant—it takes time for the condenser to reach a full charge when a voltage is applied through a resistor. The condenser voltage builds up gradually from zero, somewhat like the curve a-b-c of Fig. 31B.

Battery B_1 charges condenser C through the resistance R in Fig. 31A. If it were not for tube VT, the condenser would charge up to the battery voltage and nothing else would happen. However, tube VT is a special gas tube (like a thyratron) in which no plate current will flow until a certain critical plate voltage is applied. Then the tube suddenly begins conducting.



FIG. 31. How a sweep generator produces the desired sweep voltage.

Because of the gas, it becomes practically a short circuit.

As the condenser starts charging through R, its voltage follows the path a-b-c. When b is reached, the condenser voltage is sufficient to make VTconduct. Condenser C then discharges immediately through the tube, so the eondenser voltage drops to zero, or from b to d. This means there is no longer any plate voltage on VT, so the tube stops conducting.

Battery B_1 again tries to charge the condenser through R, along curve d-e-f. At e, VT again breaks down and the cycle is repeated. Thus the actual voltage across the condenser follows the a-b-d-e-g-h-j-k curve and has approximately the desired shape.

We want the *a-b*, *d-e*, *g-h* and j-k portions of the condenser voltage to be as straight as possible. This means the voltage source must be high enough

to make the tube VT act quickly, which will keep these points on the more nearly straight part of the condenser charging curve.

The frequency of this wave can be varied by changing the resistance of R or the capacity of C. This changes the time constant, and hence the frequency with which VT breaks down.

We can take this saw-tooth voltage off the condenser from terminals 1 and \overline{z} . A blocking condenser is usually used, feeding into a resistor, so the following circuits will not affect the shape of the sweep voltage. As you've learned, feeding a pulsating d.c. (which our sawtooth wave is) through a blocking condenser gives an a.c. input to the next circuit.

Using the Sweep Voltage. When we apply only the sweep voltage to the horizontal deflecting plates, the line in Fig. 29C is produced. The spot moves at a steady rate from left to right, then snaps back to the left to repeat the trace.

Suppose we now apply a sine wave voltage to the vertical plates, starting this wave and the sweep at the same time. Figure 32 shows what happens. The voltage wave shown at A is the sweep voltage, while B represents the voltage applied to the vertical deflecting plates. The combination of these two voltages produces the wave C on the screen.

The spot will normally be in the center of the screen, about at point e. As the sweep voltage now has an a.c. form because of the blocking condenser through which it was fed, points 1 to 9 of A represent negative potentials on the right-hand horizontal deflecting plate which corresponds to plate b of Fig. 29. Points 9 to 17 represent positive voltages on this plate (with respect to plate a).

Therefore, when the sweep starts at point 1, plate b is negative, so the spot

moves toward plate a or to the extreme left. This is point a in Fig. 32C. The effect of the changing sweep voltage is to make the right-hand plate less negative, then positive, so the spot is \bigvee moved to the right.

When the sweep voltage is at position 1, the vertical voltage is at position 2 (no voltage). The spot is therefore far over to the left at position a. When the sweep voltage increases to 3, the vertical voltage increases to 4. Now the sweep voltage moves the spot horizontally, and the vertical voltage moves it vertically at the same time.



FIG. 32. How applying the sweep and a sine wave causes the sine wave to be reproduced.

The combined effect of the two voltages moves the spot to position b, (where lines drawn from the two voltages intersect). Similarly, voltages at δ and θ move the spot to c, and so on. When the sweep voltage reaches 17, the condenser discharges, and the sweep voltage drops swiftly to point 19, then starts to build up again and repeats the whole process.

These voltages move the spot so quickly that we see a line, instead of moving spots, on the screen. The line from i to a on the screen is produced by the swift drop in sweep voltage from 17 to 19. Thus, the sweep voltage has changed the screen trace of the voltage on the vertical plates from a vertical line to a spread-out "picture" which is exactly like the applied wave. Regardless of the shape or distortion of the wave applied to the vertical plates, it will be reproduced exactly.

SYNCHRONIZATION

We started the sweep at the same time as the sine wave and had it finish at the same time. Hence, both are of the same frequency.

These frequencies must be "locked" together to prevent the cathode ray pattern from drifting about. We do this by feeding some of the input signal to the thyratron tube grid at points 3 and 4 of Fig. 31A. The sweep generator is first adjusted to a frequency just



FIG. 33. When the sweep frequency is half the vertical deflecting frequency, two cycles will be produced.

slightly below the right frequency. Then, the signals applied to the thyratron swing the grid positive, making the tube conductive and discharging the condenser at just the right moment to lock the sweep in step with the incoming wave. In other words, this forces the sweep to "jump" to the right frequency. This is called synchronization. This control will not work if the sweep frequency is too far from the right value. It is necessary to adjust the sweep frequency below the right value so the sweep won't ever discharge by itself and thus get out of step with the synchronizing grid voltage.

It is also possible to make the sweep lock at a frequency which is an exact fraction of the applied voltage frequency. This gives two or more cycles on the screen, as shown in Fig. 33. You will learn more about this in another lesson.

FREQUENCY RANGE

Very high frequencies can be applied directly to the deflection plates of a properly designed c.r.o. However, about 30 volts are needed to give a deflection of about 1 inch on the screen. This means that voltages of 75 to 100 volts or so may be required for a reasonably large figure on the screen—far higher than are usually available. For this reason, an amplifier for the input to the vertical plates is built into the c.r.o.

This is usually a resistance-coupled amplifier of from 1 to 3 stages, designed to have a wide frequency response. Even so, the amplifier limits the frequency range. Oscilloscopes made for radio service work are mostly used at audio frequencies, and so have a range up to about 100 kc. Special amplifiers are used in laboratories and in television work to gain a wide frequency range.

A volume control is used on the amplifier so that the deflection can be adjusted within the limits of the screen.

A TYPICAL C.R.O.

Figure 34 shows a block diagram of a typical oscilloscope designed for service work. Let's go through the sections starting with the c.r.o. tube.

The C.R.O. Tube. The tube is

shown here as you would find it on a schematic diagram. Starting from the filament, we have a cathode, the element G, the plate P_1 (shown here with a grid symbol), plate P_2 , the vertical deflecting plates and the horizontal deflecting plates.

The cathode connects to the junction of R_3 and R_4 . The grid element G connects to the slider of R_3 , which provides a variable negative bias used to control the intensity of the spot on the screen.

Plate P_1 connects to the slider of

of the earth.

Vertical Amplifier. The voltage being checked is fed in at the terminals V at the left. Switch SW_1 can be set to feed this signal directly to the c.r.o. through C_1 , or into the vertical amplifier. This amplifier is designed for frequencies up to 100 kc. in most service instruments. It has a volume or gain control.

Horizontal Amplifier. The signal fed to the horizontal deflecting plates may be an external signal or may come from the sweep generator. The switches



FIG. 34. A block diagram of the "innards" of a typical c.r.o.

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 R_4 , which provides a variable positive voltage, used to focus the spot accurately on the screen. Plate P_2 connects to the positive supply terminal, so that the maximum voltage difference exists between P_2 and the cathode. The power supply furnishes 1000 volts or more for the tube.

One plate of each pair of deflecting plates is connected to P_2 . The application of a signal requires a ground here, which means the positive power supply terminal is grounded instead of the negative as in radio sets. The position of the ground does not affect the power supply in any way, as it is independent SW_2 , SW_8 and SW_4 are ganged together, on the same shaft, and control the signal paths to the horizontal plates.

▶ In position 1, an external signal is fed in at the *H* terminals at the right, and passes through position 1 of switch SW_4 , then through C_2 to the c.r.o. tube. ▶ In position 2, the external signal from terminals *H* comes through SW_3 into the horizontal amplifier, and from there through SW_4 to the c.r.o. tube. This provides amplification, controlable by the gain control on this amplifier.

 \blacktriangleright Positions 3, 4 and 5 feed the sweep

generator signal through SW_3 to the horizontal amplifier and from there through SW_4 to the c.r.o. tube. These three positions provide varying sources of synchronizing voltage for the sweep generator.

▶ In position 3, part of the signal from the vertical amplifier is fed through SW_2 to the sweep input. This is the most used position of these switches, as this provides a "lock" with the incoming signal for the vertical plates.

▶ Position 4 feeds 60 cycle a.c. into the sweep generator, permitting a lock with the power line frequency.

▶ Position 5 connects the sweep to external terminals SYNC, so that an external frequency source can be used to control the sweep.

Centering Controls. It is impossible to locate the tube elements in every tube so as to get the spot in the exact center of the screen when no deflecting voltages are applied. However, this can be corrected electrically as shown in Fig. 35.

Instead of connecting P_2 and deflecting plates b and d to the maximum positive point as in Fig. 34, this connection is now made to the junction of added resistors R_6 and R_7 .

Resistors R_1 and R_2 are now grounded for a.c. through condensers C_3 and C_4 and their d.c. paths return to potentiometers R_8 and R_9 .

Resistors R_6 and R_7 are now part of the voltage divider, and the same voltage exists across R_8 and R_9 , which are in parallel. When the sliders on R_8 and R_9 are centered, the arms are at the same potential as the R_6-R_7 junction. There is now no difference in po-

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tential between the pairs of deflecting plates.

Adjusting R_8 will make plate c either positive or negative with respect to d, depending on the direction of movement from the center of the control.

Similarly R_9 will bias plate a with



FIG. 35. Centering controls correct for an offcenter spot of light.

respect to b. Thus it is possible to move the spot up or down or to the right or left, until it is exactly centered. These controls are found on commercial instruments, and are called the centering controls.

After the beam is centered and focused, and the intensity is adjusted to reasonable brilliancy, we can apply the external signals and the sweep voltage (if used) through C_1 and C_2 to the c.r.o. tube.

Lesson Questions

Be sure to number your Answer Sheet 29FR-2.

Place your Student Number on every Answer Sheet.

Send in your set of answers for this lesson immediately after you finish them, as instructed in the Study Schedule. This will give you the greatest possible benefit from our speedy personal grading service.

- 1. What ohms-per-volt sensitivity is commonly obtained on the basic range of d.c. type vacuum tube voltmeters? 1 to 5 mean hims
- 2. What is the purpose of R3 and C in Fig. 3? They art as an fr
- 3. What is the purpose of R2 in Fig. 3? Is a get bias riverton on Cathode lead. If lends to keep plate current more longe
- 4. What is the purpose of R, in Fig. 10? To get greater stability & higher sensitienty
- 5. Suppose an a.c. type v.t.v.m. is connected across a tuned circuit. How can you compensate for the capacity loading which detunes the circuit? Returne Current with VTVM in the current I make measurement
- 6. Why do we get an error when we try to measure a distorted wave with an a.c. type v.t.v.m.? a distorted wave will not have
- The same average to R. M.S. relationships as a sine the 7. What determines the accuracy of a slide-back v.t.v.m. when properly adjusted? The DC, matter MM in fig. 23
- 8. What are the advantages of using tuned circuits in signal traders? To determine the actual gain in a radio a
- 9. What is the purpose of the sawtooth voltage in a c.r.o.? To sweep the spot back I farth
- 10. When synchronizing the sweep of a c.r.o. so as to make the pattern "lock in," is the sweep frequency adjusted *above*, *below*, *or the same as* the synchronizing signal frequency?

"WISHERS" AND "DOERS"

How often have you said, "I wish I had more money?" Thousands of times, possibly. But do you realize that if you are living in a town of, let us say, 5000 inhabitants, there are exactly 4999 others in your town who are saying exactly the same thing?

And yet, of these 5000 "wishers," only about 100 are going to do something about it. The others are going to continue being "wishers."

Now, any man who shows enough "get-up-andgo" spirit to undertake this Course proves that he is not a mere "wisher." When you enrolled, you showed that you wanted to be a "go-getter." Your job now is to keep going forward on the road you have mapped out for yourself.

Every lesson in this Course, every radio job you work hard to get, is a step along this road. So don't let yourself wish that the lessons were easier, or that you could become successful without studying, or that radio jobs would come looking for you. Stay out of the class of the "wisher," and stay in the class of the "doer."

The "wisher" is a very unhappy individual because he is constantly thinking and worrying about what he does not have. The go-getter is so busy getting what he wants that he doesn't have time to be unhappy.

J. E. SMITH



NATIONAL RADIO INSTITUTE

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STUDY SCHEDULE NO. 30

For each study step, read the assigned pages first at your usual speed. Reread slowly one or more times, finish with one quick reading to fix the important facts firmly iu your mind, then answer the Lesson Questions for that step. Study each other step in this same way.

lesson, as well as the required accuracies in various radio fields.

- ment. These methods require more work and are less accurate than some of the other methods but are useful where extreme accuracy is not required. Answer Lesson Question 1.
- current flow through the coil to provide the proper core flux density. This per-mits the measurement to be made under conditions simulating actual operation. Answer Lesson Questions 2 and 3.
 - solid-dielectric and electrolytic condensers, as explained here. Answer Lesson Question 4.
- 5. Resonance Methods of Measuring L and C.....Pages 11-15 By setting up a resonant circuit, the inductance and capacity values can be accurately found. Also, the coil r. f. resistance and distributed capacity can be measured and the coil Q calculated. This method is very valuable where r. f. circuit components are being checked. Answer Lesson Questions 5, 6 and 7.
- use of bridge circuits. These circuits have the advantage of covering wide ranges, using only a few standard parts, and have a high degree of accuracy. Answer Lesson Questions 8 and 9.
- frequency standard, using a wave meter, tuning fork or reed, etc. You will find this section very interesting—giving an insight into the way service equip-ment is calibrated. Answer Lesson Question 10.
 - 8. Mail Your Answers for this Lesson to N. R. I. for Grading.
 - 9. Start Studying the Next Lesson.

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World Radio History

MEASUREMENTS AT AUDIO AND RADIO FREQUENCIES

IN past lessons you have studied inductance, capacity, impedance, reactance, power factor and the Q factor possessed by coils and condensers at audio and radio frequencies. Now we are going to learn practical ways of actually measuring these properties.

Such measurements are made by using a circuit which allows the part in question to be compared directly or indirectly with a standard part. The National Bureau of Standards, in Washington, D. C., has a set of resistors, a set of coils and a set of condensers whose resistance, inductance and capacity values are accurately known. This government agency and other large laboratories calibrate radio parts for manufacturers and others engaged in research work. These calibrated parts, in turn, are used as standards in rating ordinary radio parts.

Usually some ratio circuit is used in making comparisons, so that fewer standard parts—which are very expensive—are needed. If the ratio test circuit is put together on the workbench, some simple calculation is required, using either Ohm's Law or simple multiplication or division. However, factory-made, precalibrated capacity bridges, Q factor meters, inductance bridges and other measuring apparatus are available with directreading dials which can be read as easily as ordinary meter scales.

► The method used to make the measurements depends on the accuracy desired and on the apparatus available. If coils had only inductance, condensers only capacity and resistors only resistance, measurement would be vastly simplified. Instead, coils have distributed capacities and both a.c. and d.c. resistances; condensers have resistances and inductances; while resistors may have inductances which, though small, are quite effective at higher frequencies. Measurements must either take all these factors into account or be somewhat in error.

The accuracy with which you will wish to make measurements depends on the radio field you enter. If you become a design or research engineer. you must take into account all the properties of each radio part in your circuits. You will need to calculate carefully the values of parts to be used, then insert parts (whose values have been accurately determined) in a trial circuit to prove the design works. You will then vary parts values to find the tolerances permissible. The laboratory measuring procedures you will use require careful control and very expensive equipment.

If you enter production work in a radio factory as a laboratory technician, factory serviceman or inspector, you will need to make only semilaboratory measurements. These do not require such complex procedures nor such expensive equipment, for you will usually make comparison measurements to see if parts are alike or within tolerance limits, rather than find absolute values directly. In other words, you will compare one coil with another similar one to find their inductance difference instead of to find the actual inductance value accurately.

If you make servicing or radio communications your career, you will rarely make even these simplified measurements. Usually the radio trouble will be some simple breakdown which you can find from voltage, current or resistance measurements, and you can determine replacement parts values from the wiring diagram or previous experience with similar models. Even so, knowledge of ways of measuring coils and condensers. using ordinary shop equipment, will help you in special cases of unusual trouble and also in understanding more clearly the relations between these units.

▶ In this lesson we will discuss both simple methods and basic laboratory techniques. To do so, we here present a number of methods. Some use formulas, and in some cases we show how the formulas are obtained. We do not intend that you try to memorize the methods, the formulas, or the process of getting the formulas-they are presented here for your future reference and you can come back to this lesson should you need this information. We cover the field here, so you will have some idea of the measurements required in all branches of radio and will not have to start in absolutely green should you go into any field of radio requiring this information. You will probably never use all the methods described, no matter where you go. so just learn the basic principles of the methods and answer the lesson questions.

Let us now see how measurements are made. We'll start with air-core inductances, then take up iron-core coils, condensers, resonance methods and bridge methods.

Inductance of Air-Core Coils

Inductance values of air-core coils are rarely given on service diagrams. However, just any coil cannot be used as a replacement, particularly for an r.f. transformer, where the inductance value may determine the tuning range, and the Q factor may affect both the sensitivity and selectivity. The practical serviceman would therefore replace a defective air-core coil with an exact duplicate ordered from the receiver manufacturer or a radio supply house. If a duplicate were unobtainable, he would send the original coil to a coil manufacturer specializing in winding coils for service purposes. The serviceman is thus likely to need to find the inductance of an air-core coil only when building something for experimental purposes (a wave trap. V cannot be used when tolerance limits for instance).

To the factory man or engineer, on the other hand, inductance and Q factor are very important and must be found time and time again. Let's see how they can be measured. We'll use basic methods which are suitable for other r.f. measurements as well.

▶ The inductance of an air-core coil can be found from its physical dimensions—such as length and diameter of the coil, the number of turns and size of wire. You can then use these values with an inductance design chart or formula to determine the inductance within a few percent. These charts or formulas are found in engineering handbooks and radio publications.

However, at the time you need them. charts are not always available; they are being determined, nor do they give

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the Q factor. So now let us study actual inductance measurements. 1/ We can find the inductance of an air-core coil by determining the coil reactance and calculating the inductance, by direct comparison methods, or by using bridge or resonant circuit methods. Since the last two can also be used to measure capacity and resistance, they will be covered in another section of this lesson. Let us now take up the reactance and direct comparison methods, which use equipment found in any laboratory and in many service shops.

REACTANCE METHODS

A very simple method which can be used with reasonable accuracy on aircore coils is to apply a known a.c. voltage to the coil and determine the current flow, as in Fig. 1. From Ohm's Law for a.c. circuits, you know that the impedance of the circuit equals the voltage divided by the current flow $(Z = E \div I)$. Also, you know that the impedance Z is a combination of the a.c. resistance and the reactance of the coil. $(Z = \sqrt{R^2 + X_L^2})$. If you neglect the a.c. resistance, the impedance is the same as the reactance of the coil. Once you find the inductive reactance, you can find the inductance from the formula $X_{\rm L} = 2\pi f L$, which is easier to use if arranged in the form $L = X_L \div 2\pi f^*$.

To find the inductance of the coil by this method, you must know the frequency and neglect the a.c. resistance. The frequency will be that of the voltage source, which will probably be a signal generator when an air-core coil is being checked. For accuracy, the frequency used to determine the reactance by this method must be frequency of the coil. Since air-core coils are used in r.f. circuits, radio frequencies must be used. Usually the frequency band in which the coil is designed to work is known, so you can choose some frequency in or near this band. (Remember that if the frequency is too low, the coil will have very little reactance and an extremely high current will flow.)

somewhere near the normal working

▶ This method is quite simple, since you just measure the voltage and current and perform a calculation. How-



Courtesy General Radio Interior view of a standard variable inductance.

ever, it is only accurate enough for a very rough approximation. Remember, you neglect the a.c. resistance of the coil, which may be appreciable. Also, since you use the frequency in your calculation, any error in the calibration of the signal generator will cause a similar error in your result. Further, you must use accurate a.c. meters, because any inaccuracies in current and voltage measurements will likewise affect the final result. Only a thermocouple type current meter and an r.f. vacuum tube voltmeter are accurate enough for r.f. measurements.

▶ You must also be careful how you use your meters. In the circuit shown

^{*} L=inductance in henrys

 $X_{\rm L} =$ inductive reactance in ohms

f =frequency in cycles $2\pi = 6.28$, a constant

in Fig. 1A, the current meter $I_{\rm M}$ actually measures the current through the coil, but the voltmeter $V_{\rm M}$ measures both the voltage drop across the coil and the drop across the current meter. If the current meter has an appreciably high resistance compared to the coil reactance, the measured voltage may be considerably different from the actual voltage across the coil, as it will be the sum of the coil voltage and the voltage drop across the current meter.

You can make the voltage measurement more accurate by placing the voltmeter directly across the coil, as shown in Fig. 1B. Now, however, the current meter measures both the current through the coil and the current through the voltmeter, so the current reading will be accurate only if the voltmeter has a high ohms-per-volt sensitivity.

Which connection is better depends on the meters used. At radio frequen-



FIG. 1. Two methods of connecting meters to measure voltage and current.

cies, the current meter must be a thermocouple or hot-wire type, as other a.c. meters are usable only at power and audio frequencies. Fig. 1A can be used if the current meter has low resistance, while Fig. 1B can be used if the voltmeter has a high ohms-pervolt sensitivity. When a v.t.v.m. is used, Fig. 1B is the better circuit. ▶ Another practical difficulty with this method is that the signal generator may not have sufficient output. If so, use a single stage amplifier to increase the output sufficiently.

Obviously, with so many chances of error, this method is not extremely accurate. However, it may be good



FIG. 2. The comparison method used for an air-core coil.

enough if accuracy within 10% to 20% is permissible.

COMPARISON METHODS

If a known inductance is available, you can place the unknown inductance in series with the calibrated one, apply an r.f. voltage to the combination (as shown in Fig. 2), and get the voltage readings. The voltage drops are proportional to the impedances. If we ignore the coil a.c. resistance, then the voltages are proportional to the reactances, which in turn are proportional to the inductances, so we can say that L V_2L V_{1} - = - or $L_{\rm X} = -$. Thus, to find V_1 V_{n} $L_{\rm X}$ the inductance of $L_{\rm X}$. multiply the in-

ductance of L by the voltage V_2 and divide by the voltage V_1 . For best re-

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sults, the standard should be near the value of the unknown, so the voltmeter readings will be similar.

▶ If a variable standard inductance is available, use it as L, then adjust the variable standard inductance until the voltage drops across the two coils are equal. When they are equal, the impedances of the two coils are equal and, neglecting resistance again, you can read the unknown inductance directly from the calibrated scale on the variable standard inductance. This method has many advantages over the reactance method. For example, you do not have to know the frequency of the source-and, with the variable standard, if identical voltmeters are used across each coil or the same one is moved from coil to coil, even the accuracy of the voltmeter does not matter. However, some error may be caused by neglecting the resistances of the two coils, which may be quite different.

Resistor Comparison. If a standard inductance is not available and great accuracy is not required, the method shown in Fig. 3 can be used. Here you use a variable resistor as the variable standard and adjust it until the two voltmeters read the same value.

To make the meters read alike, you may have to adjust the frequency of the a.c. source until the coil reactance falls within the resistance range. If the voltmeter reading across the coil is higher than the reading across the resistance, even with maximum resistance, you must either lower the frequency or use a higher resistance.

Again, equal voltages mean the impedances of the two devices are equal. If you neglect inductive and capacitive effects in the resistance and also neglect resistance within the coil, you can say that the resistance of R is equal to the reactance of L_x . Then you can use the reactance formula to determine the inductance. Since the reactance of L_x is equal to R, the formula becomes $L_x = R \div 6.28f$, where L_x is in henrys, R is in ohms and f is in cycles per second.

► This method has the advantage of not requiring a standard inductance. You do not even need a calibrated resistor, because you can measure the resistance with an ohmmeter after equal voltages are obtained. However, it does require the use of either two identical voltmeters or one meter with a very high ohms-per-volt rating such as a v.t.v.m.

The accuracy is affected by neglecting both the resistance of the coil and the inductance and capacitive effects in the resistance unit, and also depends



FIG. 3. How to use a variable resistance as the standard in the comparison method.

upon the accuracy with which the resistance and frequency are determined. However, this method eliminates current measurements and does not require accurate meters, so it is rather commonly used when errors of 10% to 20% can be permitted.

Three-Voltage Method. By making three voltage measurements and using a graph, you can improve con-

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siderably the accuracy of the method of Fig. 3. The measuring circuit is shown in Fig. 4A. Here, R need not be variable, but its resistance must be known (or measured) accurately. Preferably, R should have a resistance such that $E_{\rm R}$ is somewhere near the value of the voltage $E_{\rm Z}$, but this is not necessary.

After making the three voltage measurements shown in Fig. 4A, you must draw a graph. To do this, first mark a horizontal line on a sheet of



FIG. 4. The three-voltage method. Not only is this method more accurate, it can be used when only a voltmeter is available.

paper, such as the line O-A in Fig. 4B. Along this line, measure off a distance proportional to the voltage found across resistor R. Any satisfactory scale can be used, such as 10 volts to the inch. After you have measured the proper length along this line from point O with a ruler, mark point B. The distance O-B is then proportional to the voltage $E_{\rm R}$.

Now, take a compass (a drawing instrument similar to a pair of dividers, with a pencil attachment) and adjust the compass so that the distance between its point and its pencil is proportional to the voltage E_s , using

the same scale of volts-per-inch used for laying out the line O-B. With the compass point set at point O, draw an arc or semicircle with the compass.

Next, adjust the compass so that the distance between its point and pencil is proportional to voltage E_z , using the same volts-per-inch scale as before.

With the compass point now set at point B, draw an arc with the compass that crosses the arc you made from point O. The point where they cross is point C in Fig. 4B. Now, draw lines from point C to point B and from point C to point O. The line O-C is then proportional to the voltage E_s , while the line B-C is proportional to the voltage E_z .

Next, drop a vertical line (line C-D) from point C at right angles to line O-A. You can use the right (90°) angle of a triangle to do this.

You have now formed a triangle having as its sides the lines B-C, B-Dand C-D. As line B-C represents the voltage across the coil *impedance*, the line C-D is proportional to the voltage drop across the inductive reactance alone, while the distance B-D is proportional to the voltage drop across the coil a.c. resistance. The combination of these two voltages, of course, makes up the total voltage drop across the coil, which you have found to be E_z .

Now since the voltage $E_{\rm L}$ is caused by the reactance alone, dividing this voltage by the current flow will give you the true coil reactance. The voltage $E_{\rm L}$ can be determined by measuring the line C-D with a ruler, then converting the number of inches into volts, using the same volts-per-inch scale you used in drawing the other lines.

The <u>current</u> is, of course, equal to the resistor voltage $E_{\rm R}$ divided by <u>R</u>. Compute this current, divide $E_{\rm L}$ by it, and you have the coil reactance. You can now find the true inductance from the formula $L = X_{\rm L} \div 6.28f$.

▶ This method eliminates one error by taking the coil a.c. resistance into account. Using identical voltmeters (or the same one), your accuracy depends only on the accuracy with which the resistance is found, the accuracy with which the frequency is known, and the care used in drawing the graph. Make the graph as large as possible (fewer volts-per-inch scale) for greatest accuracy.

► The three-voltage method also lets you determine the Q factor of the coil at the frequency of measurement. You know the Q factor is equal to the reactance of a coil divided by its a.c. resistance $(Q = X_L \div R)$. As the coil resistance and coil reactance are effectively in series, with the same current through each, the reactance and resistance are proportional to their voltage drops. Therefore, you can find the Q factor directly by dividing voltage $E_{\rm r}$ by voltage $E_{\rm RL}$.

Other Methods. You can also measure inductance by using various bridge circuits and resonant circuits. Since these methods can also be used to measure capacity, they will be discussed by themselves later in this lesson.

Inductance of Iron-Core Coils

You cannot determine the inductance of an iron-core coil except by electrical measurements, as its physical size does not indicate its inductance. A serviceman would probably replace a defective coil with a duplicate, but it might be necessary for him to measure the inductance to be sure the coil replacement is proper.

The methods of measuring the inductance of an iron-core device are somewhat similar to those already discussed for air-core coils. However, you must be careful to measure the inductance under the operating conditions which will exist in the radio receiver, transmitter or other apparatus, because the actual amount of inductance depends on the core flux density. If the coil is used in a circuit where d.c. flows through it, you must approximate the same d.c. flow during tests. You must also have about the same a.c. flow as in normal operation, and the frequency used for measurement must be within the normal range of the device

If the iron-core coil is used in a circuit where no d.c. normally flows, its inductance can be measured by the methods used for an air-core coil as long as a normal a.c. voltage and frequency are employed. With a transformer, you must remember that you are determining the inductance of only the one coil which is connected in the measuring circuit. The inductance of the other windings must be determined separately. Also, to find the actual operating inductance of any winding, you must connect the normal loads to all the windings of the transformereven to those not being measured.

POLARIZING CURRENT

In most radio uses, an iron-core coil will have both d.c. and a.c. flowing through it. As you know, you must send normal amounts of both a.c. and d.c. through such a coil to measure its inductance under working conditions. A circuit for doing so is shown in Fig. 5.

In this circuit, coil $L_{\mathbf{x}}$ is the induct-

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FIG. 5. Iron-core coils require a polarizing current, so this circuit is used.

ance under test. The battery B causes d.c. to flow through the coil; this current is read on I_{DC} and adjusted to the required value by resistor R.

A.C. is applied through transformer T and condenser C. Condenser C prevents d.c. from flowing through the secondary of transformer T, while choke coil L_1 prevents a.c. flow through the battery and R. Thus, d.c. flows only from battery B through R, L_1 and L_X , while a.c. flows only from the transformer through condenser C and coil L_X .

Condenser C should have a large capacity, so its reactance will be negligible. Also, L_1 must be a high-inductance coil so the a.c. flow through L_1 -R-B (in parallel with L_X) will be negligible.

After the d.c. has been adjusted to

approximate the amount which normally flows through the coil when in use, a tap on the secondary of transformer T is chosen which produces about the normal a.c. voltage drop across coil $L_{\mathbf{x}}$, as indicated by $E_{\mathbf{z}}$. (This meter must measure a.c. only, so blocking condenser C_1 is placed in series with it.) Then the current meter I_{AC} is read. Dividing the voltage E_z by the alternating current gives the impedance of the coil $L_{\rm X}$. If the a.c. resistance of the coil is assumed negligible, the impedance will be approximately the same as the reactance of the coil, and the inductance can be found from the formula $L = X_{L} \div$ 6.28f.

However, most iron-core coils are low-Q devices, having large windings with appreciable a.c. resistance which cannot be ignored. To eliminate the large error introduced when assuming the impedance equals the resistance. the best way to measure the inductance of such coils is the three-voltage method already described. To use it. first insert a known resistance at point X, then adjust the d.c. to the proper value and measure: 1, the voltage E_{s} ; 2, the voltage across the resistor at X; and 3, the voltage E_{z} . You can then use exactly the same procedure followed for Fig. 4.

Condenser Capacity

We normally deal with three kinds of radio condensers; <u>air-dielectric con-</u> densers (usually variable condensers), solid-dielectric condensers, and electrolytics.

The serviceman is interested primarily in the last two types. (Measurement of an air-dielectric condenser is almost never required in servicing, so we will cover this type later.) Next to tubes, solid-dielectric and electrolytic condensers cause more troubles than any other parts. While ohmmeter measurements or part substitution will show up most of these defects, condenser measurements are required frequently enough so that test equipment manufacturers have brought out capacity testers for servicemen. These will be described shortly, when we take up bridge circuits.

SOLID-DIELECTRIC CONDENSERS

Let us first see how to measure capacities of ordinary solid-dielectric bypass condensers—paper, mica, ceramic, and similar types. The methods we will now discuss are satisfactory for capacities above .01 mfd.

Reactance Methods. Condenser capacity can be measured in almost



Courtesy General Radio

A standard fixed condenser, which has been accurately determined and sealed within a protective container.

exactly the same manner as coil inductance, by measuring the applied a.c. voltage and the current flow. It's a wise precaution to determine first whether the condenser is short-circuited, since the alternating current meter can be damaged by excess amounts of current. Also, be sure the applied a.c. voltage peak is well below the peak voltage rating of the condenser.

After measuring the a.c. voltage and current, you can find the impedance by dividing the voltage by the current. Since the a.c. resistance of the normal solid-dielectric condenser is negligible at low frequencies, you can say that the reactance equals the impedance and then use the formula for capaci-

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tive reactance, $X_c = ----$ (where iC

f is in cycles per second and C is in microfarads), to find the capacity. This is usually converted to the han-159,000

dier form $C = -\frac{f_{c}}{f_{c}}$, where X_{c} is

the capacitive reactance in ohms.

As you know, the larger the condenser, the lower the reactance, hence the greater the current flow for a fixed voltage. This means it is possible to make a simple capacity meter with a basic circuit like that in Fig. 6. Using a fixed frequency and a fixed applied voltage, the current meter scale can be calibrated directly in terms of capacity. Shunts can be used to give different current ranges and extend the capacity range. However, the larger the condenser the lower the reactance, and hence, the higher the current. This means the condenser size range is limited by the current range and too large a condenser can cause the meter to be ruined by excess current. A shorted condenser used in this circuit will also ruin the meter. All condensers must first be checked with an ohmmeter to prevent such damage.

▶ You can also measure condenser capacity with the comparison circuit





shown in Fig. 7, which requires a vacuum tube voltmeter and a condenser C_{\bullet} of known capacity. You compare the voltages across the two condensers. As you know, the capacitive reactance increases as the capacity is decreased.
Since the voltage drop depends on the reactance, this means the larger voltage drop will appear across the smaller condenser. In other words, the voltage across each condenser is inversely proportional to its capacity.

Therefore, you can say that the ratio of the voltage across the known condenser to that across the unknown condenser equals the ratio of the capacity of the unknown to that of the



FIG. 7. The comparison method for condensers.

known condenser. Putting this in a formula, $\frac{V_s}{V_x} = \frac{C_x}{C_s}$ or $C_x = \frac{C_s V_s}{V_x}$. Thus,

to find the capacity of the unknown condenser, multiply the capacity of the known condenser by the voltage across it, then divide this product by the voltage across the unknown condenser. The answer is the capacity of the unknown condenser in the same units as the known condenser (usually in microfarads).

If C_s is adjusted so the two voltages are equal, then the capacities are equal. Thus, if you use a calibrated variable condenser at C_s , you can make the voltages equal (so C_s equals C_x) and read the value of C_x directly from the calibration of C_s .

A vacuum tube voltmeter should be used for making measurements—unless the capacities are quite large and the test frequency is lower than 10 kc., when an ordinary copper-oxide rectifier type meter can be used. If the resistance of the meter is many times higher than the reactance of the condensers, one meter will suffice. But if the meter has such low resistance that it may affect the results, two identical meters should be used.

ELECTROLYTIC CONDENSERS

The capacity of an electrolytic condenser should be measured with a d.c. voltage applied, because practically all electrolytics need a d.c. voltage to keep the film formed on the anode of the condenser. If an a.c. voltage of more than 3 or 4 volts is applied by itself, the condenser may be ruined. Also, the condenser power factor must usually be taken into consideration, since in electrolytics the series resistance is relatively high.

Fig. 8 shows a circuit for checking an electrolytic. The three-voltage method is used. As the voltmeters must read a.c. only, they are isolated by blocking condensers.

▶ First, adjust the d.c. voltage to equal some value higher than the peak of the a.c. to be applied. This value plus the applied a.c. peak value must be below the working voltage rating of the condenser. After waiting a



FIG. 8. The polarizing and testing circuit for electrolytics.

reasonable length of time for the d.c. to form the condenser, apply an a.c. voltage from the transformer. (This a.c. voltage must not exceed the d.c. voltage.) Now, measure voltages V_1 , V_2 and V_3 . Next, construct the triangle shown in Fig. 9 by the same method used in Fig. 4, using V_2 as the horizontal reference vector. Finally, drop a vertical line from the point P to form the line V_X , and extend the line V_2 to form V_R .

The line $V_{\mathbf{x}}$ represents the true voltage across the condenser reactance. Measure the length of this line, convert it into volts, and divide the voltage thus found by the actual current



FIG. 9. The three-voltage triangle for electrolytics.

flow in the circuit (which is equal to voltage V_2 divided by R in ohms). The result is the condenser reactance, which you can readily convert to capacity by using the reactance formula.

Power Factor. The power factor is found by dividing the resistance of a part by its impedance. This result is multiplied by 100 to express the power factor as a percentage.

A resistor has a power factor of 1, or 100%, because the resistance is equal to the impedance. On the other hand, a perfect coil or a perfect condenser should have zero resistance and thus zero power factor. Therefore, a coil or condenser with high power factor has considerable series resistance, and so will not act as much like a perfect inductance or perfect capacitance as will a similar device with a lower power factor.

After having made the triangles shown in Fig. 9, you can determine the power factor of the condenser by measuring the line $V_{\rm R}$ (which represents the voltage drop across the condenser resistance), then dividing this amount by the voltage V_1 . Multiply by 100 to give the power factor as a percentage. The smaller the power factor, of course, the better the condenser. Measured on 60-cycle a.c., a good electrolytic should have a power factor of about 6%. If the percentage is above 25% or 30%, you should replace the condenser.

• Another point of interest is the Q of the condenser. The larger the Q factor, the better the condenser. This factor may be obtained by dividing the value of V_x by the value of V_R . If angle θ is close to 90°, the Q factor is approximately equal to one, divided by the power factor.

▶ The circuit in Fig. 8 can be used also to measure the capacity of soliddielectric condensers. To do so, shortcircuit the d.c. power pack terminals and make measurements with a.c. only

Resonance Methods of Measuring L and C

While most of the circuits discussed so far are satisfactory for measuring power supply and audio components, they are not always usable or reliable at radio frequencies. Resonant circuits are usually better for r.f. measurements, both because they will measure small inductances and capacities accurately, and because these measurements can be made regardless of the a.c. resistance or distributed capacity of the device.

In general, to make a resonant circuit measurement, you insert the inductance or capacity to be measured in a circuit which you tune to resonance. Then, since you know the values of the other circuit components, the value of the inductance or capacity under test may be casily determined. Either a series or a parallel resonant test circuit may be used. In a series resonant circuit, a maximum current flow through the circuit or a maximum voltage across one of the parts will show when you reach resonance. In a parallel resonant circuit, a maximum voltage across the circuit, maximum voltage across the circuit, maximum circulating current, or minimum line current will all indicate resonance.

► To set up one of these measuring circuits, you need: a signal source capable of delivering about 5 watts of power, with a calibrated frequency output; some means of making measurements at radio frequencies (such as a thermocouple ammeter or an r.f. type vacuum tube voltmeter); and either a calibrated precision condenser or a known inductance. For resistance measurements, a variable resistor calibrated for r.f. will be necessary.

Fig. 10 shows two typical circuits. In Fig. 10A, you know you have reached resonance when the circulating current (measured by the thermo-





couple type meter) is at a maximum. The indicator in Fig. 10B is a v.t.v.m. which will show a maximum voltage across the condenser at resonance.

In both circuits, the coupling to the oscillator is adjusted to the minimum value which will give a reasonable meter reading. Very frequently an electrostatic shield S (shown by dotted lines) is used between the link coil and the coil in the resonant circuit to eliminate capacity coupling effects.

CAPACITY MEASUREMENTS

Suppose coil L is a known inductance. You can then measure capacity by inserting the unknown condenser in place of condenser C_s and adjusting the frequency of the oscillator until resonance is indicated by a maximum meter reading. The resistor R should be adjusted to zero resistance if the circuit in Fig. 10A is used.

Knowing the frequency and the inductance, you can determine the condenser capacity from the fact that, at resonance, the capacitive reactance equals the inductive reactance. This gives you a formula $C = \frac{25,330}{f^2L}$ (where C is the unknown capacity in microfarads, f is the frequency in kilocycles, and L is the inductance in microhenries) from which you can calculate the unknown capacity.

In this method the a.c. resistance has no effect (because you use a reactance balance, not an impedance balance), and the accuracy depends entirely upon the accuracy with which the inductance and frequency are known. Even the meter accuracy does not matter, since you use it only to indicate a maximum. However, in Fig. 10B, the input capacity of the v.t.v.m. is included in the calculated value. This capacity is small in a well-designed unit, but the measurement is in error by this amount. When doubtful about this capacity, Fig. 10A should be used if possible.

▶ If you have a calibrated condenser instead of a known inductance, you can use any coil which will give resonance and find the unknown capacity by a substitution method, provided the unknown capacity is within the range of the known capacity. First, put the unknown condenser in place of C_s and



Courtesy General Radio The front panel of a standard variable condenser.

adjust the frequency of the oscillator until resonance is indicated. Then replace the unknown condenser with the standard calibrated condenser and adjust the latter until resonance is again indicated, using the same coil and the same frequency. The capacity of the unknown condenser can now be read directly from the scale of the calibrated condenser. We do not need to / know the frequency nor the inductance value and make no calculation. Even the v.t.v.m. input capacity will not matter, as it will be present in both cases so the standard condenser has exactly the capacity of the unknown. Only the accuracy of the standard condenser determines the accuracy of the result, so this method can be very precise.

▶ Another method of using a calibrated condenser is particularly valuable where the unknown condenser is very small. Leave the calibrated condenser in the circuit and adjust the frequency of the oscillator so that resonance will be obtained with nearly maximum capacity of C_{s} . Call this C_s value C_1 . Next, shunt condenser C_{s} with the unknown capacity C_{x} , as in Fig. 10B. This increases the capacity of the circuit, so reduce the value of C_s until resonance is again obtained with the same frequency. Call this value C_2 , and find the value of the unknown condenser C_x by subtracting C_2 from C_1 . In other words, the capacity of the unknown condenser is equal to the difference in the two values of the standard condenser. If a vernier (a geared tuning arrangement or a calibrated trimmer) is used on the standard condenser $C_{\rm s}$, you can measure values as low as 2 micro-mic-, rofarads accurately by this method.

INDUCTANCE MEASUREMENTS

Either of the circuits in Fig. 10 can be used to measure inductance if you use a standard condenser and set resistor R at zero. The unknown coil is used as the inductance L. Bring the circuit to resonance, either by using a fixed capacity C_s and varying the oscillator frequency, or by leaving the oscillator frequency fixed and varying C_s , then calculate the inductance from 25,330

the formula: $L = \frac{25,555}{f^2 C}$ (where L is

in microhenries, frequency is in kilocycles and C is in microfarads).

Since the oscillator link coupling may affect the inductance somewhat, weak coupling should be used. You can easily tell when the coupling is too tight, because you will get a double resonant hump (two maximum peaks) rather than a single sharp peak indica-

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