SELECTION AND OPERATION

WIRELESS MICROPHONE SYSTEMS

By Tim Vear
Introduction

In the decades since their introduction, wireless microphone systems have evolved from a technical novelty to a universal and critical tool in productions of every type. At the same time, the ratio of performance vs. cost has steadily improved, making today’s wireless microphone an affordable and reliable choice.

As their appeal grows, wireless microphones and personal monitors are being used in larger and more complex applications. Meanwhile, the radio environment has become more crowded and challenging than ever. The ongoing proliferation of mobile wireless devices shows no sign of letting up, and as this is written, nearly 200 million smartphones are in use in the U.S. On top of this, commercial and industrial applications for wireless technology such as the ‘internet of things’ continue to grow. The result is that the radio spectrum is often a noisy, congested place. To make matters worse, available spectrum for wireless microphones is changing as regulatory agencies in many countries shuffle spectrum allocations to make enough spectrum available for everything.

Typical performance, meeting, and sports venues are now chock full of RF signals, from wireless ticket scanners to show control systems to WiFi and Bluetooth networks used by the attendees, not to mention RF noise sources like LED video walls. Even simple wireless applications require careful attention to a number of issues to insure good performance. Software for laptop computers and apps for mobile devices have made it much easier to scan the spectrum, coordinate operating frequencies for all of the wireless devices used, and monitor conditions during the event.

The scope of this guide is limited to wireless systems used primarily in audio applications. In most examples we refer to wireless microphones, but the same radio and audio principles apply to wireless in-ear monitors, wireless intercoms, and similar audio transmission systems. We presume that the reader is familiar with typical audio systems, but many of the characteristics of radio systems are significantly different and not always intuitive. For this reason, this book begins by covering some basic radio principles.

The purpose of this guide is to make it easier for you to select the right wireless system and use it successfully. We hope that the fundamentals presented here will further your understanding of this important and evolving technology.

This guide is presented in two parts: how wireless microphone systems work, and how to make wireless microphone systems work. The first part of this guide is a technical introduction to the basic principles of radio and the characteristics of transmitters and receivers (how they work). The second part discusses how to select and operate wireless microphone systems for general and specific applications (how to make them work). The first part should be of interest to those who specify or install professional wireless equipment while the second part should be of use to anyone who regularly works with wireless microphone systems.
CHAPTER ONE

BASIC RADIO PRINCIPLES

Radio Wave Transmission

Radio refers to a class of time-varying electromagnetic fields created by varying voltages and/or currents in certain physical sources. These sources may be "artificial," such as electrical power and electronic circuits, or "natural," such as the atmosphere (lightning) and stars (sunspots). The electromagnetic field variations radiate outward from the source forming a pattern called a radio wave. Thus, a radio wave is a series of electromagnetic field variations travelling through space. Although, technically, any varying source of voltage or current produces a varying field near the source, here the term "radio wave" describes field variations that propagate a significant distance away from the source.

A sound wave has only a single "field" component (air pressure). Variations in this component create a pattern of air pressure changes along the direction the sound wave travels but otherwise have no particular orientation. In contrast, a radio wave includes both an electric field component and a magnetic field component. The variations in these components have the same relative pattern along the direction the radio wave travels but they are orthogonal (oriented at a 90 degree angle) to each other as illustrated in Figure 1-1. In particular, it is the orientation of the electric field component which determines the angle of "polarization" of the radio wave. This becomes especially important in the design and operation of antennas.

Like sound waves, a radio wave can be described by its frequency and its amplitude. The frequency of a radio wave is the time rate of the field variations measured in Hertz (Hz), where 1 Hz equals 1 cycle-per-second. The radio spectrum, or range of frequencies, extends from a few Hertz through the Kilohertz (KHz) and Megahertz (MHz) ranges, to beyond the Gigahertz (GHz) range. The suffixes KHz, MHz, and GHz refer to thousands, millions, and billions of cycles-per-second respectively. As far as is presently known, humans are directly sensitive to radio waves only at frequencies in the range of a few million GHz, which are perceived as visible light, and at those frequencies in the range just below visible light, which are perceived as heat (infrared radiation). The overall radio spectrum includes both natural and artificial sources as indicated by Figure 1-2.

The amplitude of a radio wave is the magnitude of the field variations. It is the characteristic that determines the "strength" of the radio wave. Specifically, it is defined to be the amplitude of the electric field variation. It is measured in volts per unit length and ranges from nanovolts/meter (nV/m) to kilovolts/meter (KV/m), where nV refers to one billionth of a volt and KV refers to one thousand volts. The minimum level required for pickup by a typical radio receiver is only a few tens of microvolts (uV, a millionth of a volt) but much higher levels can be found near transmitters and other sources. The wide range of radio wave amplitudes that may be encountered in typical applications requires great care in the design and use of wireless microphone systems, particularly receivers.

Another characteristic of waves, related to frequency, is wavelength. The wavelength is the physical distance between the start of one cycle and the start of the next cycle as the wave moves through space. Wavelength is related to frequency by the speed at which the wave travels through a given medium. This relationship is expressed in the wave equation, which states that the speed of the wave is always equal to the product of the frequency times the wavelength. The wave equation applies to any physical wave phenomenon such as radio waves, sound waves, seismic waves, etc. (See Figure 1-3.)

The speed of radio waves (through a vacuum) is equal to approximately 3 x 10^8 meter/second, or about 186,000 miles/second. This is also known as the "speed of light," since light is just one part of the radio spectrum. The wave equation states that the frequency of a radio wave, multiplied by its wavelength always equals the speed of...
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light. Thus, the higher the radio frequency, the shorter the wavelength, and the lower the frequency, the longer the wavelength. Typical wavelengths for high-band VHF and for UHF frequencies are given in Figure 1-3. Wavelength also has important consequences for the design and use of wireless microphone systems, particularly for antennas.

Unlike sound, radio waves do not require a physical substance (such as air) for transmission. In fact, they "propagate" or travel most efficiently through the vacuum of space. However, the speed of radio waves is somewhat slower when travelling through a medium other than vacuum. For example, visible light travels more slowly through glass than through air. This effect accounts for the "refraction" or bending of light by a lens. Radio waves can also be affected by the size and composition of objects in their path. In particular, they can be reflected by metal objects if the size of the object is comparable to or greater than the wavelength of the radio wave. Large metal obstacles can reflect both low frequency (long wavelength) and high frequency (short wavelength) radio waves, but small metal obstacles can reflect only high frequency (short) radio waves. (See Figure 1-5.)

Interestingly, even a non-solid metal object can reflect radio waves. That is, the metal structure can have holes or openings in it, but as long as the openings are much smaller than the radio wavelength, the metal object will have the same effect as if it were solid. This means that screens, grids, bars, or other metal arrays can reflect radio waves whose wavelength is greater than the space between the array elements but less than the overall array size. However, if the space between elements is larger than the wavelength, the radio waves will pass through the array.

For example, the metal grid on the glass door of a microwave oven reflects microwaves back into the oven but allows light waves to pass through so that the inside is visible. Microwaves (about 2.4GHz) have a wavelength of 12.5 centimeters (about 5 inches). This wavelength is much larger than the openings in the metal grid on the oven door and thus the grid acts as if it were a solid metal wall. However, visible light has a wavelength of less than one-millionth of a meter and will pass freely through the grid openings. (See Figure 1-4)

Metal objects that are somewhat smaller than the wavelength are able to bend or "diffact" radio waves. Generally,
the size, location, and quantity of metal in the vicinity of radio waves will have significant effect on their behavior.

Non-metallic materials do not generally reflect radio waves but are not completely transparent either. To some degree, non-metallic substances generally "attenuate" or cause a loss in the strength of radio waves that pass through them. The amount of attenuation or loss is a function of the thickness and composition of the material and also a function of the radio wavelength. In practice, dense materials produce more losses than lighter materials and short radio waves (high frequencies) are attenuated more severely than long radio waves (low frequencies). The human body, for example, causes significant losses to short radio waves passing through it.

A metal object that is large enough to reflect radio waves or a non-metal object that is dense enough to attenuate them can create a "shadow" in the path of the waves which can greatly hamper reception of radio in the area beyond the object.

A final parallel between sound waves and radio waves lies in the nature of the overall radio wave pattern or "field" produced by various sources at a given receive location. In a free-field condition (no reflections) the power level of a radio wave decreases as it propagates from its source at a rate governed by the inverse-square law. This decrease is 3dB for each doubling of distance. Note that for a sound wave, sound pressure level (SPL) decreases by 6dB for each doubling of distance. The numerical difference reflects that the radio calculation is a power ratio while the acoustic calculation is a pressure ratio. For either type of wave, at a sufficient distance from the source, the wave will become too weak to detect.

If reflections are present (which is nearly always the case indoors), the radio field (like a sound field) will include both direct waves (those that travel by the shortest path from the source to the receive location) and indirect waves (those that are reflected to the receive location). Because the indirect waves travel a longer distance, they arrive at the receive location at some delay relative to the direct wave. The indirect waves can combine with the direct wave at the receive location to cause variations in the overall received field strength. This phenomenon is called "multipath" and will be discussed further in the description of radio receivers. Ultimately, the strength of a radio wave that arrives at a given location, is equal to the strength of the original source minus the amount of loss due to distance (inverse square loss), loss due to material attenuation, and variations due to multipath.

After many reflections radio waves (like sound waves) become weaker and essentially non-directional. They eventually contribute to ambient "noise," that is, general radio energy produced by many natural and man-made sources across a wide range of frequencies. The strength of ambient radio noise is relatively constant in a given area, that is, it does not diminish with distance. The total radio field at a given location consists of direct waves, indirect waves and radio noise.

Radio noise is nearly always considered to be undesirable. The direct and indirect waves may come from both the desired source (the intended transmission) and undesirable sources (other transmissions and general radio energy emitters). Successful radio reception depends on a favorable level of the desired transmission compared to the level of undesirable transmissions and noise at the specific receive location.

**Radio Wave Modulation**

This discussion of radio transmission has so far dealt only with the basic radio wave. It is also necessary to consider how information is carried by these waves. Again, a comparison of sound waves and radio waves may be useful for illustration.

Audio "information" is transmitted by sound waves which consist of air pressure variations over a large range of amplitudes and frequencies. This combination of varying amplitudes and varying frequencies creates a highly complex sound field.

Radio "information" is generally transmitted using only one frequency. This single electromagnetic wave is varied in amplitude, frequency, or some other characteristic (such as phase) and for most radio transmissions neither the wave nor its variation can be detected or processed directly by human senses. In fact, the wave itself is not the information but rather the "carrier" of the information. The information is contained in the amplitude variation or frequency variation, for example. When a radio wave contains information it is called a radio "signal." The general term for this information-carrying variation of radio waves is "modulation." If the amplitude of the wave is varied the technique is called Amplitude Modulation or AM. If the frequency is varied, it is called Frequency Modulation or FM. If the phase is varied, it is called Phase Modulation or PM.

The amount of information that can be carried in a radio signal depends on the type of modulation and the level of modulation that can be applied to the basic radio wave. Presently, both analog and digital modulation techniques are found in high quality wireless systems. Greater levels of modulation allow more information to be transmitted. The
information capacity of the radio signal is also related to the frequency of the basic radio wave. Higher frequency radio waves can generally carry more information than lower frequency waves. The information “efficiency” of both analog and digital modulation schemes is limited by physics to some extent, but may also be limited by regulatory agencies (such as the FCC in the United States).

Both analog and digital modulation utilize the same variables noted above: amplitude modulation, frequency modulation, and/or phase modulation. However, analog transmissions modulate the radio wave over a continuous range of values while digital transmissions modulate the radio wave only in discreet steps. This is the fundamental difference between the two schemes: analog modulation can have any value within the modulation limits while digital modulation can have only certain specific values within its modulation limits.

The other major difference between analog signal modulation and digital signal modulation is the relationship of the modulated carrier to the original information. An analog radio signal has the characteristics of the original information represented directly in the transmitted signal. For example, the modulation rate and modulation amount of an analog wireless microphone signal correspond proportionally to the frequency and amplitude of the original audio source.

A digital wireless microphone system necessarily includes an initial analog-to-digital conversion step (ADC) in the transmitter and a final digital-to-analog conversion (DAC) step in the receiver. In the transmitter, the ADC produces a digital data stream that is ultimately used to modulate the carrier. This data stream is recovered from the transmitted signal in the receiver and is used to reconstruct the original analog information.

The basic method of transmitting digital data in a radio signal is a step-wise modulation of the carrier called “shift keying”. As noted earlier, there are three principal radio wave parameters that can be modulated: amplitude, frequency, and/or phase. The names of the corresponding digital modulation techniques are ASK (amplitude shift key), FSK (frequency shift key), and PSK (phase shift key). It is also possible to combine multiple digital modulation techniques to transmit more information.

For AM signals, the radio wave has a single (constant) frequency of some basic amplitude (determined by the transmitter power). This amplitude is varied up and down (modulated) by the transmitting signal to create the corresponding radio signal.

In an analog AM signal, the rate of modulation is equal to the frequency of the audio signal and the amount of modulation is proportional to the amplitude (loudness) of the audio signal. Primarily due to regulatory limits, an analog AM transmission typically allows an audio signal of only limited frequency response (about 50-9000 Hz) and limited dynamic range (about 50 dB). (See Figure 1-6a.)

A digital AM signal (ASK), modulates the amplitude of the radio wave in discrete steps. At a minimum, the ASK modulation can have two values, corresponding to two amplitude values. A data bit is represented by a shift from one amplitude value to the other. In a very simplified example, a shift from the lower amplitude to the higher amplitude might represent binary “0” while a shift from the higher amplitude to the lower amplitude might represent binary “1”. The data is transmitted as a series of amplitude shifts representing the bit values. (See Figure 1-6b)

For analog FM signals, the radio wave has a constant amplitude (again determined by transmitter power) and a basic frequency. The basic radio frequency is varied up and
down (modulated) by the audio signal to create the corresponding radio signal. This frequency modulation is called "deviation" since it causes the carrier to deviate up and down from its basic or unmodulated frequency. (See Figure 1-7a.)

As with an analog AM signal, the rate of modulation is equal to the audio frequency and the amount of modulation is proportional to the audio amplitude. Typical values of deviation in analog FM wireless microphone systems range from about +/- 12 KHz to +/- 45 KHz depending on the operating frequency band. The maximum (legal) amount of deviation allows an audio signal of greater frequency response (about 50-15,000 Hz) and greater dynamic range (more than 90 dB) than does analog AM.

A digital FM signal (FSK) modulates the frequency of the radio wave in discrete steps. Again, at a minimum, the FSK modulation can have two values corresponding to two frequency values. A data bit is represented by a shift from one frequency value to the other. Similar to the simplified example of the ASK technique, a shift from the lower frequency to the higher frequency might represent binary "0" while a shift from the higher frequency to the lower might represent binary "1". The data is transmitted as a series of frequency shifts representing the bit values. (See Figure 1-7b)

A corresponding comparison can be made between analog phase modulation (PM) and digital phase shift key modulation (PSK).

A consideration that must be noted in any discussion of digital audio or digital radio systems is "latency". Due to the finite processing time of digital circuitry, there is a measurable delay between an input signal and the corresponding output signal. In a real-time audio application this delay (latency) must be kept as short as possible to minimize audible artifacts that may distract from the performance.

Although the details of wireless microphone transmitters and receivers will be covered in the next section, it should be noted here that all of the analog wireless systems discussed in this presentation use the FM technique. The reasons for this are the same as are apparent in commercial analog broadcast systems. More "information" can be sent in the typical analog FM signal, allowing higher fidelity audio signals to be transmitted. In addition, analog FM receivers are inherently less sensitive to many common sources of radio noise, such as lightning and electrical power equipment. These sources are characterized by a high level of AM-type noise which is rejected by analog FM systems. Finally, the audio quality of an analog FM signal is not directly dependent on signal strength. Even if the received signal strength fluctuates substantially due to distance or multipath, there will no audible effect until the strength falls close to the RF background noise level.

The digital wireless systems presented subsequently make use of ASK, FSK, PSK, and combination modulation techniques. The choice of digital modulation scheme for a particular application is certainly related to the required audio quality (including minimizing latency) but it may also be dictated by the requirements of a chosen radio frequency band among other factors.

![Figure 1-7a: frequency modulation (FM)](image1)

![Figure 1-7b: frequency shift key (FSK)](image2)
CHAPTER TWO

BASIC RADIO SYSTEMS

System Description

The function of a radio or "wireless" system is to send information in the form of a radio signal. In this presentation, the information is assumed to be an audio signal, but of course video, data, or control signals can all be sent via radio waves. In each case, the information must be converted to a radio signal, transmitted, received, and converted back to its original form. The initial conversion consists of using the original information to create a radio signal by "modulating" a basic radio wave. In the final conversion, a complementary technique is used to "demodulate" the radio signal to recover the original information.

A wireless audio system consists generally of three main components: an input source, a transmitter, and a receiver. (See Figure 2-1.) The input source provides an audio signal to the transmitter. The transmitter converts the audio signal to a radio signal and "broadcasts" or transmits it to the surrounding area. The receiver "picks up" or receives the radio signal and converts it back into an audio signal. Additional system components include antennas and, possibly, antenna cables and distribution systems. The processes and the basic components are functionally similar to commercial radio and television and other forms of radio communications. What differs is the component scale and the physical system configurations.

There are four basic configurations of wireless audio systems, related to the mobility of the transmitter and receiver components, as required for different applications.

The first configuration involves a portable transmitter and a stationary receiver. The transmitter is usually carried by the user, who is free to move about, while the receiver is located in a fixed position. The input source in this setup is normally a microphone or an electronic musical instrument. The receiver output is typically sent to a sound system, recording equipment, or a broadcast system. This is the configuration of the standard "wireless microphone" and is the arrangement most widely used in entertainment, public address, and broadcast applications.

The second configuration employs a stationary transmitter and a portable receiver. In this case, the user carries the receiver, while the transmitter is fixed. The input source to the transmitter for these setups is usually a sound system, playback system, or other installed source. The output of the receiver is typically monitored through headphones or loudspeakers, though it may feed a portable audio or video recorder. This is the configuration of wireless systems for in-ear-monitors (IEMs), interruptible foldback systems (IFB), assistive listening, simultaneous translation, and various instructional uses. It is also, of course, the configuration of commercial radio and television broadcast systems when the receiver is mobile such as a personal radio or a car radio.

The third configuration consists of both a portable transmitter and a portable receiver. The users of both components are free to move about. Again, the input source is usually a microphone and the output is often a headphone. This is the configuration of "wireless intercom" systems, though each user in a typical setup has both a transmitter and a receiver for two-way communication. Another application of this configuration is for transmission of audio from a wireless microphone to a portable camera/recorder in broadcast, film, and videography.

The fourth configuration comprises a transmitter and a receiver that are each stationary. Such setups are often referred to as "point-to-point" wireless systems. The typical input would be a playback source or mixer while the output might be to a sound system or to a broadcast facility. Examples of this setup are wireless audio feeds to multiple amplifier/loudspeaker arrays for temporary distributed sound systems, radio remote-to-studio links and of course commercial and non-commercial broadcasts from fixed transmitters to fixed receivers.

Analog, Hybrid, and Digital Systems

As other analog links in the audio chain have moved into the digital domain, wireless audio systems have also adopted digital technology. Digital techniques are employed in professional wireless in several ways, each offering potential benefits as well as actual challenges.

The first level of application has been the use of digital control circuits for various tasks such as frequency selection, diversity antenna switching and display functions. Most current analog wireless systems benefit from the extensive use of non-audio digital circuitry.

The next digital application level employs DSP (Digital Signal Processing) to replace traditional analog processing circuits. An audio DSP circuit is used in the transmitter to
optimize the input signal for transmission and a complementary audio DSP is used in the receiver to optimize the output signal. Because the radio transmission path is still in the analog domain, these are often called “hybrid” systems.

The highest level of digital implementation uses a fully digital transmission path. The input signal is digitized in the transmitter and can remain in the digital domain until the receiver output, or even deliver a digital signal from the receiver to subsequent digital equipment.

Input Sources
The input source is any device that provides a suitable audio signal to the transmitter. "Suitable audio signal" means an electrical signal within a certain frequency range (audio), voltage range (microphone level or line level), and impedance range (low or high) that can be handled by the transmitter. Though this places some limits on input sources, it will be seen that almost any type of audio signal can be used with one system or another.

The most common input source is a microphone, which may take any one of a variety of forms: handheld, lavalier, headworn, instrument-mounted, etc. The audio signal provided by this source is audio frequency, microphone level, and usually low impedance. Since the "wireless" part of the wireless microphone only serves to replace the cable, ideally, the characteristics and performance of a particular microphone should not change when used as part of a wireless microphone system. Therefore, the selection of microphone type for a wireless microphone system should be made following the same guidelines as for wired microphones. The usual choices of operating principle (dynamic/condenser), frequency response (flat/shaped), directionality (omnidirectional/unidirectional), electrical output (balanced/unbalanced, low or high impedance), and physical design (size, shape, mounting, etc.) must still be made correctly. Problems that result from improper microphone choice will only be aggravated in a wireless application.

Another widely encountered input source is an electronic musical instrument, such as an electric guitar, electric bass, or portable electronic keyboard. The signal from these sources is again audio frequency, microphone or line level, and usually high impedance. The potentially higher signal levels and high impedances can affect transmitter choice and operation.

Finally, general audio signal sources such as mixer outputs, portable music players, laptop computers, etc. may be considered. These exhibit a wide range of levels and impedances. However, as long as these characteristics are within the input capabilities of the transmitter they may be successfully used.

Transmitter: General Description
Transmitters can be either fixed or portable as mentioned earlier. Regardless of type, transmitters usually feature a single audio input (line or microphone type), various controls and indicators and a single antenna. Internally, they are also functionally the same, except for the power supply: AC power for fixed types and battery power for portable models. The important features of transmitter design will be presented in the context of portable units.

Portable transmitters are available in three principal forms: bodypack, handheld, and plug-on. (See Figure 2-2.) Each of these has further variations of inputs, controls, indicators, and antennas. The choice of transmitter type is often dictated by the choice of input source: handheld microphones usually require handheld or plug-on transmitters while nearly all other sources are used with bodypack types.

Bodypack (sometimes called beltpack) transmitters are typically packaged in a shirt-pocket sized rectangular housing. They are often provided with a clip that secures the transmitter to clothing or a belt, or may be placed in a pocket or pouch. In theater and some other applications they may be concealed underneath clothing. Input is made from the source to the bodypack via a cable, which may be permanently attached or detachable at a connector. This connector may allow a variety of input sources to be used with one transmitter.

Bodypack transmitter controls include at least a power switch and sometimes a separate mute switch, allowing the audio input to be silenced without interrupting the radio signal. Other controls may include gain adjustment, attenuators, limiters and of course a provision for transmitter frequency selection. Indicators for power-on and battery condition are typical, as well as some form of frequency indication. A few transmitters are equipped with audio "peak" indicators. High performance transmitters may feature menu-driven displays that offer customization of the transmitter operation. Finally, the antenna for a bodypack transmitter may be in

![Figure 2-2: examples of transmitters (left to right: handheld, bodypack, plug-on, gooseneck, boundary)
the form of a flexible attached wire, a detachable short "rubber ducky" type, or the input source cable itself, such as a guitar cable or lavaliere microphone cable. In some models the antenna is internal to the unit.

Handheld transmitters, as the name implies, consist of a handheld vocal microphone element integrated with a transmitter built into the handle. The complete package appears only slightly larger than a wired handheld microphone. It may be carried in the hand or mounted on a microphone stand using an appropriate swivel adapter. Input from the microphone element is direct via an internal connector or wires. Some models have removable or interchangeable microphone elements. Handheld transmitter controls typically include power, gain adjustment, frequency selection and sometimes a mute function. Indicators are comparable to those in bodypack transmitters: power status, battery condition, frequency. Again, high performance models may feature menu-driven displays. Handheld transmitter antennas are usually concealed internally, though certain types (primarily UHF) may use a short external antenna.

"Plug-on" transmitters are a special type designed to attach directly to a typical handheld microphone, effectively allowing many standard microphones to become "wireless." The transmitter is contained in a small rectangular or cylindrical housing with an integral female XLR-type input connector. Controls and indicators are comparable to those found in bodypack types and the antenna is usually internal.

Two additional transmitter types are those integrated with a gooseneck microphone or with a boundary microphone. These are frequently found in conference and meeting room applications when it is not practical to use traditional wired versions. The transmitter circuitry is the same as is used in bodypack types but the antenna is internal. Again, the microphone characteristics are dictated by the application requirements.

While transmitters vary considerably in their external appearance, internally they all must accomplish the same task: use the input audio signal to modulate a radio carrier and transmit the resulting radio signal effectively. Though there are many different ways to engineer wireless transmitters, certain functional elements are common to most current designs, whether analog or digital. It is useful to describe these elements to gain some insight to the overall performance and use of wireless audio systems. (See Figure 2-3.) The similarities and differences between analog and digital approaches will be noted where appropriate.

Transmitter: Audio Circuitry

The first part of the typical transmitter is the input circuitry. This section makes the proper electrical match between the input source and the rest of the transmitter. It must handle the expected range of input levels and present the correct impedance to the source. Gain controls and impedance switches allow greater flexibility in some designs. In certain cases, the input circuitry also provides an electrical voltage to the source (for condenser microphone elements). Note that this "bias" voltage (typically about 5vdc) is only intended to operate the impedance converter in a miniature condenser element. It is not the same as phantom power that is used to operate a complete condenser microphone. However, some plug-on transmitters are capable of providing phantom power to use with standard design condenser microphones.

The signal from the input stage passes to the signal processing section, which optimizes the audio signal in several ways for the constraints imposed by radio transmission. While the input stage is generally an analog...
since this technique most closely tracks the amplitude of the average signal level. A "true RMS" detector is preferred, especially when the radio signal itself is weak or noisy. The performance of full-band companding systems can control the noise intrusion that accompanies low-level audio signals, but opposite (1:2) "expansion" in the receiver, the original dynamic range of the audio signal is restored. A voltage-controlled-amplifier (VCA) is the circuit element that provides both dynamic functions: gain is decreased in the compressor mode and increased in the expander mode. The gain change is proportional to the signal level change. Nearly all current analog wireless audio systems employ some form of companding, allowing a potential dynamic range greater than 100 dB. (See Figures 2-4 a & b.)

The second process in a fully analog transmitter is called "companding" (compress/expand), which is designed to compensate for the limited dynamic range of analog radio transmission. The part of the process performed by the transmitter is "compression," in which the dynamic range of the audio signal is reduced or compressed, typically by a fixed ratio of 2:1. Again, when this is coupled with an equal but opposite (1:2) "expansion" in the receiver, the original dynamic range of the audio signal is restored. A voltage-controlled-amplifier (VCA) is the circuit element that provides both dynamic functions: gain is decreased in the compressor mode and increased in the expander mode. The gain change is proportional to the signal level change. Nearly all current analog wireless audio systems employ some form of companding, allowing a potential dynamic range greater than 100 dB. (See Figure 2-5a.)

A limitation of fixed-ratio companders is that the same amount of signal processing is applied regardless of signal level. Dynamics processors perform compression or expansion functions based on an evaluation of the average signal level, which fluctuates continuously. Because this process is not instantaneous, the compander action is not completely transparent. With good design, audible artifacts are minimal but may become more apparent when the signal level is extremely low. This accounts for occasional "modulation" noise or background noise intrusion that accompanies low-level audio signals, especially when the radio signal itself is weak or noisy. The performance of full-band companding systems can be improved by first optimizing the measurement of the average signal level. A "true RMS" detector is preferred, since this technique most closely tracks the amplitude of a full range audio signal, regardless of frequency response. Further improvement can be realized by using level-dependent companding. For low level audio signals, little or no processing is applied so there are no audible effects. As the audio signal level increases, processing levels are increased, so that potentially audible artifacts are masked. Implementation of this scheme requires a high performance VCA and close tolerance in the audio sections of transmitters and receivers. (See Figure 2-5b.)

A variation that is found in a few compander designs is to divide the audio signal into two or more frequency bands. Each band is then pre-emphasized and compressed independently. In the receiver, de-emphasis and expansion are applied separately to these same bands before combining them back into a full-range audio signal. Though more expensive, multi-band companding systems may have better ability to improve dynamic range and apparent signal-to-noise ratio across the entire audio range.

In many transmitters, an additional process called limiting is applied to the audio signal. This is to prevent overload and distortion in subsequent audio stages or to prevent "overmodulation" (excessive frequency deviation) of the FM radio signal. The "limiter" automatically prevents the audio signal level from exceeding some preset maximum level and is usually applied after pre-emphasis and companding.

In a hybrid transmitter, the audio processing described above is implemented using digital signal processing (DSP). These "hybrid" systems transmit an analog RF signal but the audio path in the transmitter and/or receiver passes through DSP circuitry. DSP can overcome many of the limitations of analog signal processing noted above, resulting in even better frequency response, wider dynamic range, and lower distortion than a fully analog system.

A fully digital transmitter does not attempt to duplicate any of these analog processes. Rather, the analog signal from the input stage is passed directly to an analog-to-digital converter (ADC). The resulting digital data stream passes to a DSP that performs multiple steps. The digital data is processed into a stream of digital "words" that contain the coded audio as well as timing and error correction values. Additional data may include battery level, audio gain settings, and other transmitter information. This processed data stream is then passed to the radio transmitter section.
Transmitter: Radio Circuitry

In a fully analog transmitter or in a hybrid transmitter, the processed audio signal is sent to a voltage-controlled oscillator (VCO). This is the section that actually converts the audio signal to a radio signal by the technique called frequency modulation (FM). The (relatively) low frequency audio signal controls a high frequency oscillator to produce a radio signal whose frequency "modulates" or varies in direct proportion to the audio signal. The modulation rate is equal to the audio frequency and the modulation amount is equal to the audio amplitude. The maximum value of modulation is called the deviation and is usually specified in kilohertz (KHz). (See Figures 2-6 a & b.)

Though an un-modulated carrier occupies only a “single” frequency, the modulated carrier occupies a range of frequencies defined by the deviation of the signal. If the deviation is +/- 45 kHz, then the minimum occupied bandwidth of the signal is 90 kHz. Allowing for some operating area around the FM signal the effective occupied bandwidth is approximately twice the deviation or about 180 kHz. Ultimately, the maximum allowable bandwidth of an FM signal is restricted by regulations in each specific radio frequency band. In the US for example, the maximum occupied bandwidth of an FM signal in the UHF range is less than 200 kHz.

The amount of deviation produced by the audio signal is a function of the design of the transmitter. Systems with deviation greater than the modulating frequency are called wideband, while systems with deviation less than the modulating frequency are called narrow band. Typical deviation for FM wireless audio systems is +/- 45 kHz. Since the highest modulating frequency is perhaps 20 kHz, most high performance analog wireless audio transmitters fall into the lower end of the wideband category. Narrowband systems are primarily used in intercom, IFB and speech-only applications. The smaller occupied bandwidth of the narrowband design allows a greater number of simultaneous systems with some reduction in audio fidelity.

The “base” or unmodulated frequency of the VCO for most tunable systems is adjustable by a technique known as frequency synthesis. A control circuit called a phase-locked-loop (PLL) is used to calibrate the transmitter frequency to a reference "clock" frequency through an adjustable frequency divider. By changing the divider in discrete steps, the transmitter base frequency can be precisely varied or tuned over the desired range. Frequency-synthesized designs allow the audio signal to modulate the VCO directly at the transmitter frequency. (See Figure 2-7.)

For a digital transmitter, the data stream must pass through a DAC (digital-to-analog converter) to provide a series of discrete control voltages to the modulating circuit of the transmitter. The transmitter itself may produce ASK, FSK, PSK or other discrete value output as needed and it is again tunable by means of a frequency-synthesized VCO. (See Figure 2-8.)

![Figure 2-6a: unmodulated FM signal spectrum](image)

![Figure 2-6b: modulated FM signal spectrum](image)

![Figure 2-7: frequency-synthesized transmitter](image)

![Figure 2-8: frequency-synthesized digital transmitter](image)
As noted earlier, the principal difference between the output of an analog (or hybrid) transmitter and a digital transmitter is in the nature of the modulation. The analog modulation is continuous within its limits while the digital modulation has only discrete values within its limits. A consequence of this difference is that an analog signal is only fully modulated when an audio signal is present while a digital signal is fully modulated whether audio is present or not. Thus the occupied bandwidth of a digital signal is constant while the occupied bandwidth of an analog signal varies with the amplitude of the analog signal.

The final stage of both analog and digital transmitters is a radio frequency (RF) power amplifier. Typical output power values may range from about 1mW up to 250mW, though this is usually restricted by regulatory agencies. It is often dependent on the modulation scheme and on the frequency band of operation.

The last internal element of the transmitter is the power supply. For portable transmitters, power is generally supplied by batteries. Since the voltage level of batteries falls as they are discharged, it is necessary to design the device to operate over a wide range of voltage or to employ voltage-regulating circuitry. Designs requiring a 9 V battery typically use the battery voltage directly. Transmitters using 1.5 V cells use DC-to-DC converters that maintain constant internal voltage levels by increasing battery current as the battery voltage decreases. This technique yields stable operation during the entire battery discharge cycle and permits very efficient battery operation. Units equipped with such voltage regulation must have a suitable protection circuit to avoid battery damage due to excess current draw. In addition, battery run-time indicators can be made more accurate since they monitor both current and voltage throughout operation. In practice, battery life varies widely among transmitters, from just a few hours up to perhaps twenty hours, depending on output power, battery type, and overall circuit efficiency.

**Receiver: General Description**

Receivers are available in both fixed and portable designs. (See Figure 2-9.) Portable receivers resemble portable transmitters externally: they are characterized by small size, one or two outputs (microphone/line, headphones), minimal controls and indicators (power, level), and (usually) a single antenna. Internally they are functionally similar to fixed receivers, again with the exception of the power supply (battery vs. AC). The important features of receivers will be presented in the context of fixed units, which exhibit a greater range of choices.

Fixed receivers offer various outward features: units may be free standing or rack-mountable; outputs may include balanced/unbalanced microphone or line level as well as headphones; indicators for power and audio/radio signal level may be present; controls for power and output level are usually offered; antennas may be removable or permanently attached. Like transmitters, receivers can vary greatly in packaging, but inside they must achieve a common goal: receive the radio signal efficiently and convert it into a suitable audio signal output. Once again it will be useful to look at the main functional elements of the typical receiver. Differences between analog, hybrid, and digital receivers will be noted where appropriate. (See Figure 2-10.)

**Receiver: Radio Circuitry**

The first section of receiver circuitry is the “front end.” Its function is to provide a first stage of radio frequency (RF) filtering to prevent unwanted radio signals from causing interference in subsequent stages. It should effectively reject signals that are substantially above or below the operating frequency of the receiver. For a single frequency receiver the front end can be fairly narrow. For a tunable receiver it must be wide enough to accommodate the desired range of frequencies if the front end filter itself is not tunable.

In order to accommodate very wide tuning ranges, a track-tuned front-end filter is sometimes used. This is a relatively narrow filter that is automatically centered on the tuned frequency to prevent excessive off-channel radio energy from entering the receiver. (See Figure 2-11.)
The second receiver section is the "local oscillator" (usually abbreviated as "LO"). This circuit generates an unmodulated radio frequency that is related to the frequency of the received radio signal but differs by some "defined amount." A typical value of this "defined amount" is 10.7 MHz. For example, if the desired receiver frequency is 200 MHz, the required LO frequency would be 210.7 MHz. A tunable (frequency-agile) receiver has an adjustable LO, which generally uses a frequency synthesis design. (See Figure 2-12a.)

The receiver is tuned to a desired frequency by tuning the LO to the desired frequency plus the "defined amount". In the above example, in order to tune the receiver to a frequency of 180 MHz, the LO would be set to 190.7 MHz (180 MHz + 10.7 MHz).

Next, the (filtered) received signal and the local oscillator output are input to the "mixer" section. The mixer, in a radio receiver, is a circuit that combines these signals by a process called "heterodyning." This process produces multiple output signals: one signal is at a frequency that is the sum of the received signal frequency and the local oscillator frequency, while another is at a frequency that is the difference between the received signal frequency and the local oscillator frequency. Additionally, integral multiples of the LO frequency and the received signal frequency are generated in the mixer. Note that in the heterodyning process, any output signal that is related to the received (modulated) signal also contains the modulation information. Specifically, both the sum and the difference signals contain the audio information carried by the received signal.

All of the signals produced by the mixer are then sent to a band-pass filter stage that is tuned to the frequency of the difference signal. This frequency is the "intermediate frequency" (IF), so-called because it is lower than the received radio frequency but still higher than the final audio frequency. It is also the "defined amount" used to determine the local oscillator frequency of the previous section. The narrowly tuned IF filters are designed to completely reject the sum signal, as well as the LO frequency (and its multiples), the original received signal (and its multiples), and any other higher frequency radio signals that may have passed through the front end filter. The IF filter allows only the difference signal to pass through. (See Figure 2-13.)

Since the IF signal still contains the modulation information, this effectively converts the received radio frequency (RF) signal to the much lower intermediate frequency (IF) signal and makes subsequent signal processing more efficient. This overall process is called "downconversion."

If only one LO and one mixer stage are used then only one intermediate frequency is produced and the receiver is said to be a "single conversion" type. In a "double conversion" receiver the incoming signal is converted to the final IF in two successive stages, each with its own LO...
and mixer. This technique can provide increased stability and interference rejection, though at significantly higher design complexity and cost. Double conversion is more common in UHF receiver designs where the received signal frequency is extremely high. (See Figure 2-12b.)

It should be noted that the LO frequency can be above or below the received frequency and still yield the same difference frequency when combined in the mixer. When the LO frequency is lower than the received frequency the design is called "low-side injection." When it is above it is called "high-side injection."

In an analog (or hybrid) receiver, the IF signal is finally input to the "detector" stage which "demodulates" or extracts the audio signal by one of several methods. One standard technique is known as "quadrature." When two signals are out of phase with each other by exactly 90 degrees they are said to be in quadrature. When such signals are multiplied together and low-pass filtered the resulting output signal consists only of the frequency variations of the original input signal.

This effectively eliminates the (high-frequency) carrier frequency leaving only the low-frequency modulation information (the original audio signal).

In an analog quadrature FM detector the IF signal passes through a circuit which introduces a 90 degree phase shift relative to the original IF signal. The phase-shifted IF signal is then multiplied by the straight IF signal. A low-pass filter is applied to the product, which results in a signal that is now the audio signal originally used to modulate the carrier in the transmitter.

In a digital receiver, the demodulation of the IF signal is generally accomplished in the digital domain. The resulting information is FSK, ASK, PSK or other discrete value output. (See Figure 2-14.)

**Receiver: Audio Circuitry**

The demodulated audio signal still exhibits the results of the signal processing that was done in the transmitter. Regardless of whether the transmitter processing was fully analog, hybrid, or fully digital the receiver must provide appropriate complementary signal processing to accurately recover the original signal. For fully analog systems, a fixed (or signal-dependent) expansion is applied, followed by a high-frequency de-emphasis. If a multi-band process was used in the transmitter, the received audio is divided into the corresponding bands, each band is expanded, the high frequency band is de-emphasized, and finally the bands are recombined to yield the full-range audio signal.

A hybrid receiver may use digital signal processing of the audio signal to complement the processing (analog or digital) that was done in the transmitter.

A fully digital receiver also implements a “complementary” process that reconstructs the original audio signal. This process begins with stream of discrete values coming from the demodulating step. This data stream is eventually passed through a digital-to-analog converter (DAC) to re-generate the original audio signal.

Finally, an output amplifier supplies the necessary audio signal characteristics (level and impedance) for connection to an external device such as a mixer input, a recorder, headphones, etc. Typically, better receivers will include a balanced output that can be switched between line level and microphone level. Unbalanced outputs are usually provided as well. Note that many digital receivers (and some hybrid receivers) are also able to output a digital signal directly to subsequent devices such as digital mixers.

**Receiver: Squelch**

One additional circuit that is important to proper receiver behavior is called "squelch" or muting. The function of this circuit is to mute or silence the audio output of the receiver in the absence of the desired radio signal. When the desired signal is lost (due to multi-path dropout, excessive distance, loss of power to the transmitter, etc.) the "open" receiver may pick up another signal or background radio "noise." In analog systems, this may be heard as 'white' noise and is often much louder than the audio signal from the desired source.
The traditional squelch circuit is an audio switch controlled by the radio signal level using a fixed or manually adjustable threshold (level). (See Figure 2-15.) When the received signal strength falls below this level, the output of the receiver is muted. Ideally, the squelch level should be set just above the background radio noise level or at the point where the desired signal is becoming too noisy to be acceptable. Higher settings of squelch level require higher received signal strength to unmute the receiver. Since received signal strength decreases as transmission distance increases, higher squelch settings will decrease the operating range of the system. One refinement of the standard squelch circuit is referred to as "noise squelch." (See Figure 2-16.) This technique relies on the fact that the audio from undesirable radio noise has a great deal of high frequency energy compared to a typical audio signal. The noise squelch circuit compares the high frequency energy of the received signal to a reference voltage set by the squelch adjustment.

In this system, the squelch control essentially determines the "quality" of signal (signal-to-noise ratio) required to unmute the receiver. This allows operation at lower squelch settings with no likelihood of noise if the desired signal is lost.

A further refinement is known as "tone-key" or "tone-code" squelch. It enables the receiver to identify the desired radio signal by means of a supra- or sub-audible tone that is generated in the transmitter and sent along with the normal audio signal. The receiver will unmute only when it picks up a radio signal of adequate strength and also detects the presence of the tone-key. This effectively prevents the possibility of noise from the receiver when the desired transmitter signal is lost, even in the presence of a (non-tone-key) interfering signal at the same frequency.

Turn-on and turn-off delays are incorporated in the transmitter tone-key circuits so that the transmitter power switch operates silently. When the transmitter is switched on, the radio signal is activated immediately but the tone-key is briefly delayed, keeping the receiver muted until the signal is stable. This masks any turn-on noise. When the transmitter is switched off, the tone-key is deactivated instantly, muting the receiver, but actual turn-off of the transmitted signal is delayed slightly. This masks any turnoff noise. As a result, the need for a separate mute switch is eliminated. Finally, the tone-key signal is often used to transmit additional information to the receiver. This may include battery voltage, transmitter audio gain settings, transmitter type, and transmitter power level. (See Figure 2-17.)

In a fully digital wireless system, the squelch process is not directly related to the RF signal level, or to the signal-to-noise ratio or even to a specific tone-key frequency. Rather, the squelch circuit is activated whenever the error rate of the detected digital data stream exceeds some preset threshold that could cause an audible artifact. This error rate dependence takes into account most of the audio and radio factors that can degrade the signal quality and can generally respond more quickly and accurately than the standard squelch mechanisms of analog or hybrid systems.

Receiver: Antenna Configuration

Fixed receivers are offered in two basic external configurations: diversity and non-diversity. Non-diversity receivers are equipped with a single antenna while diversity receivers generally have two antennas. Both systems may offer otherwise similar outward features: units may be free standing or rack-mountable; outputs may include balanced/unbalanced microphone or line level as well as headphones; indicators for power and audio/radio signal level may be present; controls for power and audio output level are provided; antenna(s) may be removable or permanently attached. (See Figure 2-18.)

Though diversity receivers tend to include more features than non-diversity types, the choice of diversity vs. non-diversity receiver is usually dictated by performance and reliability considerations. Diversity receivers can significantly improve both qualities by minimizing the effect of variations in radio signal strength in a given reception area due to fading or due to multi-path. Fading is a loss of signal strength at excessive distance or because of shadowing or blocking of the radio wave. Multi-path is a more complex phenomenon but both mechanisms can adversely affect radio reception.
Multipath

A necessary element in the concept of diversity radio reception is the occurrence of "multi-path" effects in radio transmission. In the simplest case, radio waves proceed directly from the transmitting antenna to the receiving antenna in a straight line. The received signal strength is only a function of the transmitter power and the distance between the transmitting and receiving antennas. In practice, this situation could only occur outdoors on level, unobstructed terrain.

In most situations, however, there are objects that attenuate radio waves and objects that reflect them. Since both the transmitting and receiving antennas are essentially omnidirectional, the receiving antenna picks up a varying combination of direct and reflected radio waves. The reflected waves and direct waves travel different distances (paths) to arrive at the receiving antenna, hence the term multi-path. (See Figure 2-19.)

In an analog (or hybrid) receiver, the audible effects of such signal strength variation range from a slight swishing sound ("noise-up"), to severe noises ("hits"), to complete loss of audio ("dropout"). Similar effects are sometimes noted in automobile radio reception in areas with many tall buildings. The "size" of a dropout region is related to wavelength: in the VHF range (long wavelength) dropout areas are larger but farther apart, while in the UHF range (short wavelength) they are smaller but closer together. For this reason, multi-path effects tend to be more severe in the UHF range. These effects are unpredictable, uncomfortable, and ultimately unavoidable with single-antenna (non-diversity) receivers.

Receiver: Diversity Techniques

Diversity refers to the general principle of using multiple (usually two) antennas to take advantage of the very low probability of simultaneous dropouts at two different antenna locations. "Different" means that the signals are statistically independent at each location. This is also sometimes called "space diversity," referring to the space between the antennas.

For radio waves, this "de-correlation" is a function of wavelength: a separation of one wavelength results in nearly complete de-correlation. In most cases, at least one-quarter wavelength separation between antennas is necessary for significant diversity effect: about 40 cm for VHF systems and about 10 cm for UHF systems. Some increased benefit may be realized by increasing separation to at least one wavelength. Spacing beyond three or four wavelengths does not significantly improve diversity performance, but large or unusually shaped areas may be covered with somewhat greater antenna separation.

There are a number of diversity techniques that have had some degree of success. The term "true" diversity has come to imply those systems which have two receiver sections, but technically, any system which samples the radio field at two (or more) different locations, and can "intelligently" select or combine the resulting signals is a true diversity system.

The simplest technique, called "passive antenna combining" utilizes a single receiver with a passive combination of two or three antennas. Antennas combined in this manner create an "array," which is essentially a single antenna with fixed directional characteristic. In its most effective form (three antennas, each at right angles to the other two) it can avoid complete dropouts, but with a reduction of maximum range. This is because the array output will almost always be less than the output of a single antenna at the optimum location. If only two antennas are used, dropouts can still occur in the event of an out-of-phase condition between them. Cost is relatively low but setup of multiple antennas can be somewhat cumbersome. This is not a "true" diversity design. (See Figure 2-21.)

A diversity variation of this technique is "antenna phase diversity." It also employs two antennas and a single receiver but provides an active combining circuit for the two antennas. This circuit can switch the phase of one antenna relative to the other, eliminating the possibility of
phase cancellation between them. However, switching noise is possible as well as other audible effects if switching is incorrect. Range is sometimes greater with favorable antenna combinations. Cost is relatively low. Setup requires somewhat greater antenna spacing for best results. (See Figure 2-22.)

The next variation, “antenna switching diversity,” again consists of a single receiver with two antennas. The receiver includes circuitry that selects the antenna with the better signal according to an evaluation of the radio signal. Switching noise is possible but this system avoids the possibility of phase cancellation between antennas because the antennas are never combined. Range is the same as for a single antenna system. Cost is relatively low and setup is convenient. (See Figure 2-23.)

In both of these active antenna diversity approaches, the switching decision is based on the received signal quality of a single receiver section. When the signal quality falls below some preset threshold, switching occurs immediately. If the new antenna (or antenna combination) doesn’t improve the reception, the receiver must switch back to the original state. The lack of “predictive” ability often causes unnecessary switching, increasing the chance of noise. The switching speed is also critical: too fast and audible noise occurs, too slow and a dropout may occur.

A variation of the antenna switching method offers predictive diversity capability using a microcontroller to optimize switching characteristics. A running average signal level and a maximum signal level are calculated by analyzing the change in signal level over time. Comparing the current average signal level to the most recent maximum signal level determines the switch action, based on typical dropout characteristics. Small declines at high signal levels indicate impending dropout, causing a switch to occur. At moderate signal levels, larger decreases are allowed before switching. At very low signal levels switching is curtailed to avoid unnecessary noise. Of course, if the signal level is increasing, no switching occurs. The onset of dropout can be more accurately recognized and countered, while eliminating switching when there is little likelihood for improvement.

”Receiver switching diversity” is a widely used diversity system. It consists of two complete receiver sections, each with its own associated antenna, and circuitry that selects the audio from the receiver that has the better signal. Switching noise is possible but when properly designed these systems can have very good dropout protection with little chance of other audible effects due to incorrect selection. This is because the system continuously compares the signal condition at each receiver output before switching occurs. Range is the same as with single antenna systems. Cost is higher, but setup is convenient. (See Figure 2-24.)
“Ratio combining diversity” also uses two complete receiver sections with associated antennas. This design takes advantage of the fact that, most of the time, the signal at both antennas is useable. The diversity circuitry combines the outputs of the two receiver sections by proportionally mixing them rather than switching between them. At any given moment, the combination is proportional to the signal quality of each receiver. The output will usually consist of a mix of the two audio sections. In the case of loss of reception at one antenna, the output is chosen from the other section. Excellent dropout protection is obtained with no possibility of switching noise since the diversity circuit is essentially an intelligent panpot, not a switch. (See Figure 2-25.) Signal-to-noise is improved by up to 3 dB. Range can be greater than with single antenna systems. Cost is somewhat higher, setup is convenient.

A properly implemented diversity system can yield measurable improvements in reliability, range, and signal-to-noise ratio. This is particularly true if the RF environment is severe (multipath), troubleshooting time is minimal (no rehearsal), or dropout-free performance is required (ideally always). The majority of wireless microphone systems today, both analog and digital, employ diversity designs at nearly all performance levels.

In some fully digital systems, it is possible to actively combine the bit streams from multiple antenna/receiver sections to compensate for data losses in individual bit streams. As long as sufficient data can be assembled from this combination, reception can be maintained even in the event of significant data losses in an individual channel. Some products can combine up to four individual antenna/receiver sections in this manner to create a four-way digital diversity system. Such a system can provide reliable audio even in reception conditions where an analog system would likely fail.

An additional method for improving signal reliability in the presence of interference is called “Frequency Diversity”. This technique relies on the low likelihood of simultaneous interference on two different radio frequencies. To set up a frequency diversity system requires two transmitters, each set to a different frequency, and two matching receiver channels. The two received signals are connected to an audio mixer on two separate channels. If the signal from either transmitter is interrupted, the audio engineer can continue with the remaining signal.

Presently, this is practical only by using two bodypack transmitters on a single individual, typically with lavaliere or headworn microphones. They may be connected to a single microphone or possibly two closely-spaced microphones. Frequency diversity is generally reserved for the primary user in critical situations where the cost of “double-packing” is justified. However, handheld transmitters are now available that can transmit simultaneously on two different frequencies. In addition, the matching receivers can automatically transition between the two signals when interference occurs so that only a single mixer channel is required and no manual intervention is necessary to maintain signal continuity.

**Antennas**

In addition to the circuitry contained inside transmitters and receivers, one critical circuitry element is often located outside the unit: the antenna. An antenna is electrically “resonant” (tuned) at a particular frequency or range of frequencies. Since frequency is inversely related to wavelength (via the Wave Equation), the size of the antenna is directly related to the corresponding radio wavelength(s). Lower radio frequencies require larger antennas, while higher frequencies use smaller antennas. Another characteristic of antennas is their relative efficiency at converting electrical power into radiated power and vice versa. An increase of 6 dB in radiated power, or an increase of 6 dB in received signal strength can correspond to a 50% increase in range. Likewise, a loss of 6 dB in signal may result in 50% decrease in range. Finally, antennas are said to be “reciprocal”. That is, they can generally operate in a transmitting mode or in a receiving mode with exactly the same characteristics. The exception is an “active” (amplified) antenna which can be used only in a receiving mode due to its built-in receive-only amplifier.

The function of an antenna is to act as the interface between the internal circuitry of the transmitter (or receiver) and the external radio signal. In the case of the transmitter, it must radiate the desired signal as efficiently as possible, that is, at the desired strength and in the desired direction. Since the output power of most transmitters is limited by regulatory agencies to some maximum level, and since battery life is a function of power output, antenna efficiency is critical. At the same time, size and portability of transmitters is usually very important. This results in only a few suitable designs for transmitter antennas. (See Figure 2-26.)
The smallest simple antenna that is consistent with reasonable transmitter output is an antenna that is physically (and electrically) one quarter as long as the wavelength of the radio wave frequency being transmitted. This is called a "1/4 wave" antenna. It takes different forms depending on the type of transmitter being used. For some bodypack transmitters, the antenna is a trailing wire cut to an appropriate length. In other designs the cable that attaches the microphone to the transmitter may be used as the antenna. In either case, the antenna must be allowed to extend to its proper length for maximum efficiency. The effective bandwidth of this antenna type is great enough that only about three different lengths are required to cover the UHF range. For transmitter applications requiring even smaller antenna size a short "rubber duckie" antenna is sometimes used. This type is still (electrically) a 1/4 wave antenna, but it is wound in a helical coil to yield a shorter package. There is some loss in efficiency due to the smaller "aperture" or physical length. In addition, these antennas have a narrower bandwidth.

Handheld transmitters generally conceal the antenna inside the body of the unit, or use the outer metal parts of the case as the antenna. In either design, the antenna is rarely a true 1/4 wave long. This results in somewhat less radiated power for a handheld transmitter with an internal antenna than a comparable bodypack design with an external antenna. However, antenna output is somewhat reduced when placed close to the body of the user. Since the antenna of a hand-held transmitter is usually at some distance from the body, though, the practical difference may be small. Plug-on type transmitters normally use the microphone body and the transmitter case itself as the antenna, though some manufacturers models have used an external antenna. In practice the typical VHF transmitter antenna is less than 10% efficient. UHF types may be significantly better because the shorter wavelength of these frequencies is more consistent with the requirement for a small antenna.

In all of these designs, the radio wave pattern emitted by the antenna is omnidirectional in the plane perpendicular to the axis of the antenna. For a vertically oriented antenna the radiation pattern is omnidirectional in the horizontal plane, which is the typical case for a trailing wire antenna. There is very little output along the axis of the antenna. A three-dimensional representation of the field strength from a vertical antenna would resemble a horizontal doughnut shape with the antenna passing through the center of the hole. (See Figure 2-27.)

Recall that a radio wave has both an electric field component and a magnetic field component. A vertically oriented transmitter antenna radiates an electric field component that is also vertical (while the magnetic field component is horizontal). This is said to be a "vertically polarized" wave. Horizontal orientation of the antenna produces a "horizontally polarized" wave.

In receiver applications, the antenna must pick up the desired radio signal as efficiently as possible. Since the strength of the received signal is always far less than that of the transmitted signal this requires that the antenna be very sensitive to the desired signal and in the desired direction. However, since the size and location of the receiver are less restrictive, and since directional pickup may be useful, a much greater selection of antenna types is generally available for receivers.

Again, the minimum size for adequate reception is 1/4 wavelength. A whip or telescoping antenna of this size is supplied with most receivers, and it too is omnidirectional in the horizontal plane when it is vertically oriented. An important consideration in the performance of a 1/4 wave receiving antenna is that its efficiency depends to some extent on the presence of a "ground plane," that is, a metal surface at least 1/4 wave long in one or both dimensions and electrically connected to the receiver ground at the base of the antenna. Typically, the receiver chassis or receiver PC board to which the antenna is attached acts as a sufficient ground plane. (See Figure 2-28.)

If more sensitivity is desired, or if it is necessary to mount an omnidirectional antenna remotely from the receiver, 1/2 wave or 5/8 wave antennas are often used. These antennas have a theoretical "gain" (increase of sensitivity) up to 3 dB greater than the 1/4 wave antenna in some configurations. This can translate into increased range for the system. However, the 5/8 wave antenna, like the 1/4 wave type, only achieves its performance with an appropriate ground plane. Without a ground plane unpredictable effects may occur.
resulting in asymmetric pickup patterns and potential signal loss due to the non-ideal cable/antenna interface. A properly designed 1/2 wave antenna does not require a ground plane, allowing it to be remotely mounted with relative ease. It can also maintain proper impedance at the cable/antenna interface or can be directly attached to a receiver or antenna distribution system. In addition, it is resistant to the effects of electrical noise that might otherwise be picked up at the interface.

Both 1/4 wave and 1/2 wave antennas are omnidirectional in the horizontal plane when mounted vertically. In a transmit mode, a vertically oriented antenna produces a vertically polarized radio wave. In the receive mode, a vertically oriented antenna is most sensitive to a vertically polarized radio wave. For maximum efficiency, transmit and receive antennas should be oriented in the same direction. Efficiency is reduced when the receive antenna is rotated away from the polarization angle of the incoming wave. Minimum efficiency occurs when the receive antenna is at 90 degrees relative to the wave polarization angle. This condition is called cross-polarization and may result in signal reduction of up to about 12dB in some cases. Because the polarization angle of a radio wave changes with the orientation of the transmit antenna and with reflection, it is inevitable that there will be some fluctuation in received signal strength when the transmitter orientation is changed. Using a diversity receiver, it is recommended to orient the two receiving antennas at a 90 degree angle to each other. This insures that at least one of the receiving antennas is never more than 45 degrees off from the polarization angle of the incoming radio wave.

Unidirectional antennas are also available for wireless microphone systems. These designs are comprised of a horizontal boom with multiple transverse elements and are of the same general type as long range antennas for television reception. They can achieve high gain (up to 10 dB compared to the 1/2 wave type) in one direction and can also reject interfering sources coming from other directions by as much as 30 dB. (See Figure 2-29.)

Two common types are the Yagi and the log-periodic. The Yagi consists of a dipole element and one or more additional elements: those located at the rear of the boom are larger than the dipole element and reflect the signal back to the dipole while those located at the front are smaller than the dipole and act to direct the signal on to the dipole. The Yagi has excellent directivity but has a fairly narrow bandwidth and is usually tuned to cover just one TV channel (6 MHz). The log-periodic achieves greater bandwidth than the Yagi by using multiple dipole elements in its array. The size and spacing between the dipoles varies in a logarithmic progression so that at any given frequency one or more dipoles are active while the others are functioning as reflecting or directing elements, depending on their size and location relative to the active element(s). The longer the boom and the greater the number of elements the greater is the bandwidth and the directivity. A typical log-periodic antenna has 6-8 dB of forward gain and a horizontal coverage angle of about 120 degrees. (See Figure 2-30.) Because both the Yagi antenna and the log periodic antenna are constructed from dipole elements, they exhibit the same polarization angle properties as a single dipole. For example, when oriented vertically, they transmit (or receive) vertically polarized radio waves.

The helical antenna is a special type of directional antenna. It is a broadband antenna with up to 12 dB of forward gain but with a conical coverage angle of only about 60 degrees. However, the helical antenna does not have a preferred polarization angle. That is, when used as a receive antenna it is equally sensitive to incoming signals polarized at any angle. Likewise, when used as a transmit antenna, the radiated signal can be picked up equally well by receive antennas oriented at any angle. This makes the helical antenna particularly useful when a high-gain transmit antenna is needed for use with non-diversity receivers such as in-ear monitor packs.

Although these directional antennas are somewhat large (3-5 ft. wide for VHF) and may be mechanically cumbersome to mount, they can provide increased range and greater rejection of interfering sources for certain applications. It should also be noted here that these antennas should be oriented with the transverse elements in the vertical direction rather than the horizontal direction (as would be used for television reception), again because the transmitting antennas are usually also vertical.

![Figure 2-29: examples of remote receiver antennas](image)

![Figure 2-30: log periodic antenna polar pattern (viewed from above, pointing to right)](image)
Antenna Cable

An important but often overlooked component of many wireless microphone systems is the antenna cable. Applications in which the receiver is located away from the transmitter vicinity and/or within metal racks will require the use of remote antennas and connecting cables. Compared to audio frequency signals, the nature of radio frequency signal propagation in cables is such that significant losses can occur in relatively short lengths of cable. The loss is a function of the cable type and the frequency of the signal. Figures 2-31 and 2-32 give some approximate losses for various commonly used antenna cables at different radio frequencies. It may be noted from this chart that these cables have a "characteristic" impedance, typically 50 ohms. Ideally, for minimum signal loss in antenna systems, all components should have the same impedance: that is the antennas, cables, connectors and the inputs of the receivers. In practice, the actual losses due to impedance mismatches in wireless receiver antenna systems are negligible compared to the losses due to antenna cable length. For this reason, it may be acceptable to substitute certain 75 ohm cable types when the specified loss-per-unit-length at a given frequency is comparable to the corresponding 50 ohm type. For example, a quality RG-6 (75 ohm) cable may be substituted for a similar RG-8/RG-213 (50 ohm). Note that 75 ohm connectors are required for 75 ohm cable, but 75 ohm BNC connectors will mate with 50 ohm BNC connectors.

Obviously, the benefits of even a high gain antenna can be quickly lost using the wrong cable or too long a cable. In general, antenna cable lengths should be kept as short as possible. Antenna amplifiers can be used to compensate for losses in long cable runs. (See Figure 2-36.) It is recommended that the net loss in the antenna system be no more than 5dB.

In addition, the construction of the cable should be considered: coaxial cables with a solid center conductor and stiff insulator/shield are most suitable for permanent installation, while cables with stranded conductors and flexible insulator/ shield should be used for portable applications which require repeated setups. Finally, the number of connections in the antenna signal path should be kept to a minimum.

Antenna Distribution

The last component found in larger wireless receiver systems is some form of antenna signal distribution. It is often desirable to reduce the total number of antennas in multiple systems by distributing the signal from one set of antennas to several receivers. This is usually done to simplify system setup, but can also improve performance by reducing certain types of interference as will be seen later. There are two general types of antenna distribution available: passive and active. Passive antenna splitting is accomplished with simple in-line devices that provide RF impedance matching for minimum loss. Still, a single passive split results in about a 3 dB loss, which may translate into some loss of range. (See Figure 2-34.) Multiple passive splits are impractical due to excessive signal loss.

To allow coupling of antenna signals to more receivers and to overcome the loss of passive splitters, active antenna distribution amplifiers are used. These are also known as "active antenna splitters" or "antenna multi-couplers." These devices provide enough amplification to make up for splitter loss, and usually operate at “unity” gain overall, that is, no net amplification occurs. Though a multi-coupler is generally a separate accessory, some receiver designs are equipped with internal antenna distribution when multiple receiver sections are incorporated in the same chassis.

Stand-alone active antenna splitters can typically feed up to four receivers from one set of antennas. (See Figure 2-35a.) If more receivers are required, the outputs of one distribution amplifier can feed the inputs of a second level set of distribution amplifiers. (See Figure 2-35b.) Each of these can then feed several receivers. Two or possibly three levels of active splitters is the maximum recommended, due to the potential for increased RF distortion and interference.

Certain receiver designs incorporate active antenna "cascade" outputs. Since these outputs are unity gain, it is possible to cascade the antenna signals from one receiver to the next, up to as many as ten units. A limitation of the typical "daisy chain" scheme is that all receivers must be in the same frequency band. Another consideration is that in the event of power loss or antenna connection loss to a particular receiver, all subsequent receivers in the cascade will lose antenna signal. Nevertheless, active cascade ports can simplify hookup and reduce the requirement for stand-alone active splitters.

<table>
<thead>
<tr>
<th>Type Of Cable</th>
<th>Loss @ 200MHz (100 ft)</th>
<th>Loss @ 800MHz (100 ft)</th>
</tr>
</thead>
<tbody>
<tr>
<td>RG58C/U (Belden 8262)</td>
<td>7.5 dB</td>
<td>18.5 dB</td>
</tr>
<tr>
<td>RG9/U (Belden 9559)</td>
<td>4.5 dB</td>
<td>12 dB</td>
</tr>
<tr>
<td>R0213U (Belden 8207)</td>
<td>2.7 dB</td>
<td>7.1 dB</td>
</tr>
<tr>
<td>RG8/U (Belden 9013)</td>
<td>1.8 dB</td>
<td>3.9 dB</td>
</tr>
</tbody>
</table>

Figure 2-31: comparison of coaxial cable types

Figure 2-32: coaxial antenna cable loss at VHF and UHF frequencies
Large Room/Multi-Room Applications

Sometimes it is desired to use a single wireless transmitter throughout a very large space or in multiple rooms. It is difficult to get reliable reception from transmitters in distant rooms or in extremely large rooms, especially if there are many obstructions or strong RF interference. A centrally located receiver antenna may improve the situation. Line-of-sight transmitter to receiver placement is always the preferred setup.

If a diversity receiver with detachable antennas is used the two antennas may be located in different rooms, though this essentially reduces the receiver mode to two non-diversity sections. If diversity reception is to be maintained two antennas may be located in each room with the use of an antenna combiner. The "A" antenna in one room is combined with the "A" antenna in the other room using an antenna combiner. The "B" antennas are similarly connected and the "A" and "B" combiner outputs are fed to the receiver "A" and "B" antenna inputs. (See Figure 2-33.)

It is also possible to use multiple receivers and antennas tuned to the frequency of a single transmitter. The audio outputs of the receivers can then be combined in a mixer to allow continuous pickup of the signal from multiple locations. However, some type of audio level control must be employed since the audio level of such a system will increase by 3dB each time the number of active receivers doubles. That is, if the transmitter is picked up by two receivers at the same time the overall audio level will be 3dB louder than when picked up by only one receiver. Automatic mixers can control this effect.
CHAPTER THREE

WIRELESS SYSTEM OPERATION

Frequency Bands For Wireless Systems

Existing wireless microphone systems transmit and receive on a specific radio frequency, called the operating frequency. Individual radio frequencies are found in frequency “bands” which are specific ranges of frequencies.

Use of radio frequencies in the United States is regulated by the FCC (Federal Communications Commission). The FCC has designated certain bands of frequencies and certain frequencies in those bands for use by wireless microphones, as well as by other services. In the US, the frequencies used for wireless audio systems may be grouped into six general bands or ranges:

- Low-band VHF (49–108 MHz)
- High-band VHF (169–216 MHz)
- Low-band UHF (450–698 MHz)
- High-band UHF (900–952 MHz)
- 1.9 GHz range (1920–1930 MHz)
- 2.4 GHz range (2400–2483.5 MHz)

The FCC continues to investigate other bands for potential wireless microphone use as well. Note that VHF stands for “Very High Frequency,” and UHF stands for “Ultra High Frequency.” (See Figure 3-1.)

The FCC further determines who can operate in each band and who has priority if more than one user is operating. “Primary” users include licensed broadcasters (radio and television) and commercial communications services (2-way radio, pagers, and cellular telephones). Wireless microphone operators are always considered to be “secondary” users. In general, priority is given to primary users: secondary users may not interfere with primary users, and secondary users may be subject to interference from primary users.

On the subject of licensing, it should be noted that while manufacturers must be licensed by the FCC to sell wireless equipment, it is the responsibility of the operator to observe FCC regulations regarding their actual use.

We will briefly describe each band and its advantages and disadvantages for wireless microphone system operation, based on the designated users of the band, the physical characteristics of the band, and the regulatory limitations of the band.

The VHF Band

At the beginning of the low-band VHF range is the 49 MHz region, used not only by wireless microphones but also by cordless telephones, walkie-talkies, and radio controlled toys. 54–72 MHz is occupied by VHF television channels 2–4. The 72 MHz area is used by “assistive listening” type wireless microphone systems (ALS). 76–88 MHz is assigned to VHF television channels 5 and 6. At the top, 88–108 MHz is the commercial FM radio broadcast band. (See Figure 3-3.) All of these regions have been used at one time or another for wireless microphone systems. Allowable deviation limits (typically up to 15 kHz) can accommodate high-fidelity audio (the same as for FM broadcast). The propagation of these waves through the air is very good, as is their ability to pass through many non-metallic substances (a result of their relatively long wavelength). The most attractive feature of operation in this band was low equipment cost.

Except for assistive listening systems, however, low-band VHF is not recommended for serious applications. Due to the large number of primary and secondary users, and high levels of general radio frequency (RF) “noise,” this band is prone to interference from many sources. Transmitter power is limited to less than 50 mw (except in the 72–76 MHz range where up to 1 watt is allowed for assistive listening systems). Finally, the minimum proper antenna size for units in this range can be over one meter long (one quarter of a five meter wave), which can severely limit portability and/or efficiency.

Next is the high-band VHF range, historically used for professional applications, in which quality systems were available at a variety of prices. In the US, the high-band VHF range is divided into two bands, which are available to wireless microphone users. The first of these, from 169–172 MHz, includes eight specific frequencies designated by the FCC (Part 90.263b or just “Part 90”) for wireless microphone use by general business. These frequencies are often referred to as “travelling frequencies,” because they can (theoretically) be used throughout the US without concern for interference from broadcast television. Legal limits of deviation (+12 kHz) allow high quality audio transmission. Once again, power is limited to 50 mw. Propagation characteristics are good, and antenna length is more manageable at about one-half meter for a quarter-wave type.
Operation of Wireless Systems Outside of the U.S.

Allocation and regulation of radio frequencies is supervised by specific government agencies in each country, with the result that allowable (legal) frequencies and frequency bands differ from country to country. In addition to frequency, these agencies typically specify other aspects of the equipment itself. They include: allowable transmitter power, maximum deviation (for FM), spurious emissions, etc. These specifications may differ from one band to another and may further differ from one type of user to another within the same band. For this reason, it is not possible to select a specific frequency or even frequency band that is (legally) useable in all parts of the world. Furthermore, it is not possible to design a single type of wireless equipment that will satisfy the specifications of all or even most of these agencies around the globe.

Figure 3-2: international wireless frequencies
Unfortunately, the primary users in this band include many business band and government operations such as digital paging services, forestry, hydro-electric power stations, and the Coast Guard. Since the secondary user category is not restrictive, the potential for interference from both primary and other secondary users is always present. Also, general RF noise is still fairly strong in this band. In addition, due to the limitation of available frequency bandwidth, and the spacing of the prescribed eight frequencies, it is only feasible to operate, at most, two or three units simultaneously on travelling frequencies. Finally, these frequencies are not generally legal outside of the US and Canada.

The larger part of the high-band VHF region is 174-216 MHz. This band is designated by the FCC for both licensed and unlicensed wireless microphone use. The primary users of this band are TV stations assigned to VHF channels 7-13. During the transition from analog to DTV broadcasting, many TV stations moved to the UHF band, leaving the VHF band largely vacant in many cities and therefore quite viable for wireless microphone operation. It is possible that some DTV stations may move back into the VHF band following the FCC’s auction of spectrum in the UHF band, but this remains to be seen. Once again, high quality audio is possible within legal deviation limits (+/- 15 kHz). Transmit power restrictions are the same (50 mW) for unlicensed users, but licensed users may operate at up to 250 mW. Propagation losses are still minimal, and acceptable quarter-wave antenna sizes range down to less than one-half meter.

The possibility of interference from other secondary users and general RF noise exists, but it is much less likely than for low-band frequencies. In addition, although this range includes powerful primary users (television channels 7-13), there are ample frequencies available (locally unused television channels) in almost any part of the US.

**Licensing:** The FCC has updated the licensing rules for wireless audio system users in the VHF and UHF TV bands. Unlicensed users may operate with a transmit power up to 50 mW. Licensed users may operate at power levels up to 250 mW, and are entitled to protection from interference from TV White Space Devices by registering their use of wireless microphones at specific times and locations in one of the TV White Space Databases. Qualification for an FCC wireless microphone license is limited to parties who regularly use more than 50 wireless microphones, personal monitors, and related devices. In addition to broadcasters and motion picture/video production companies, this may allow some large theaters and AV rental and staging companies to obtain a license.
The UHF Band

Low-Band UHF

The low-band UHF range of frequencies may be considered as two overlapping bands: low (450-512 MHz) and high (470-698 MHz). The primary users of these bands are business services such as land mobile radio and pagers (450-512 MHz), public safety (470-512 MHz), and UHF TV channels 14-51 (470-698 MHz). As in the high-band VHF region, unused television channels are available for wireless microphones and other wireless audio devices. Most cities have a significant number of TV stations operating in the range, and the largest cities may also have public safety operations on selected channels between 14 and 20. Though RF noise is lower in the UHF band, and there are usually sufficient unused television channels presently, this spectrum is becoming increasingly crowded.

Like high-band VHF, users may be licensed or unlicensed in the UHF band. The required minimum quarter-wave antenna size for UHF radio waves is 9-16 cm (only one-quarter to one-third that for VHF). Equipment is available at a wide range of prices and nearly all are frequency-agile diversity systems, and many are digital. Very high quality audio can now be achieved as well as a large number of simultaneous systems.

In early 2017, the FCC completed an auction of spectrum in the UHF TV band. After a 39-month transition period ending in July 2020, the “new” UHF TV band will extend up to 614 MHz, and the 614-698 MHz range will mostly be allocated to mobile broadband providers.

Wireless microphone operations will continue to be permitted on a licensed or unlicensed basis up to 608 MHz. (Television channel 37, occupying 608-614 MHz, is reserved for use by Radio Astronomy, Wireless Medical Telemetry Systems, and White Space Devices.) Wireless microphone operations will be permitted with some restrictions in the Guard Band (614-616 MHz) that serves as a buffer between the TV band and mobile broadband spectrum, and in the Duplex Gap (653-663 MHz) that separates the uplink and downlink portions of the mobile broadband spectrum. Transmit power is limited to 20 mW in these sections, and unlicensed users are prohibited
Selection and Operation of WIRELESS Microphone Systems

from operating in the 653-657 MHz range. Given the relatively small size of these frequency ranges, their proximity to higher power mobile broadband signals in the adjacent bands, and the operating restrictions, it is unclear how practical these sections will be for typical wireless microphone users.

Although the UHF television band has been reduced in size by regulatory agencies in the U.S. and elsewhere, it remains the largest contiguous block of spectrum in most countries – particularly in the U.S. At the same time, UHF wireless systems have become considerably more spectrally efficient and can now fit almost three times as many microphones into one broadcast TV channel as was possible several years ago. For this reason, the UHF band will likely remain the primary range for professional wireless audio use.

High-Band UHF

The high-band UHF range (900-1000 MHz) includes some sections that are open to unlicensed users and some that require a license. The 902-928 MHz band originally was popular for consumer products such as cordless telephones, but those devices have largely moved to much higher portions of the spectrum. This band is open for unlicensed use, and there are several high quality digital wireless microphone systems available that operate in this range. The primary limitation for most of the 900 MHz equipment is that a maximum of about five systems may be operated simultaneously in this range due to occupied bandwidth requirements.

Another portion of the high-band UHF range that may be noteworthy (for licensed users only) is the 941-960 MHz range. This consists of the 944-952 MHz range that has long been reserved for links between broadcast studios and remote transmitters (usually located on a nearby mountain or tall building), plus parts of the sections immediately above and below that range which are used by a variety of radio communication services. Specifically, these ranges are:

- 941.500-952.000 MHz
- 952.850-956.250 MHz
- 956.450-959.850 MHz

Given the non-contiguous nature of these sections of spectrum, their availability to licensed users only, and the presence of other services, the 941-960 MHz range is of marginal utility to most wireless audio users, and very little equipment that operates in these bands is currently available.

Above 1 GHz

The spectrum above 1 GHz was for many years the realm of industrial, scientific, medical, microwave, or satellite equipment that used radio waves for either short-range sensing or long-range point-to-point data links. As technology has improved and regulations have evolved, however, these frequency ranges have become practical for wireless audio systems as well.

While not available in the U.S., spectrum in the 1.5 GHz and 1.8 GHz range is legal for wireless audio system operation in parts of Europe or Asia. Professional equipment now available offers a viable alternative for users in these regions, where spectrum for wireless microphones is often much more constrained than in the U.S.

The 1.9 GHz band (1920-1930 MHz in most of the Americas, and somewhat larger in Europe) has emerged as a new option for some wireless audio applications. Regulations for this band are based on the use of the “DECT” transmission scheme originally developed for telephone applications. Modern circuitry designs are able to deliver professional audio quality and a reasonable amount of latency. Initial applications of these products have focused on videoconferencing in meeting venues, where a small amount of latency is of little concern since video compression and transmission over the network takes considerably longer.

The 2.4 GHz band (2400-2483.5 MHz) has also attracted interest for wireless microphone use. Potential advantages for this band are: very short antennas (less than 4 cm), no high power primary users, almost international availability, no bandwidth/modulation scheme limits, and no license requirement. Unfortunately, these same features have attracted a considerable number of users, including cordless telephones, short- and medium-range wireless network applications like Bluetooth and Wi-Fi, and other users. However, many of the devices in this band use a “spread spectrum” modulation scheme which reduces mutual interference to some extent. Some manufacturers have developed digital wireless microphones that include intelligent interference detection and avoidance technology that allows them to successfully operate in the presence of Wi-Fi networks and other 2.4 GHz equipment.

As wireless technology continues to evolve and the demand for spectrum increases, it is likely that we will see wireless audio systems operating in even higher frequency ranges.
Frequency Selection

Selecting the operating frequency of a wireless audio system is a two-step process: first, choose an appropriate radio frequency band; second, choose the appropriate operating frequency (or frequencies) within that band. As indicated above, there is a finite number of wireless microphone systems that may be used simultaneously in any one frequency range. The reasons for this limitation are several and they fall under the general topic of frequency coordination or "compatibility." We will define these factors and examine each in terms of origin, effects, and relative importance to total system compatibility. In the first section we will consider only interactions that may occur between the individual wireless microphone systems themselves. Outside (non-system) interactions will be discussed in the second section.

At this point it should be stated that frequency coordination is a process that must take into account various factors including the physical (limitations), the mathematical (calculations) and the political (regulations). It is not necessary for most users to go through the details of this process. Wireless manufacturers provide this service through pre-selected frequency groups and can also offer assistance in complex applications using proprietary computer programs. What follows here is an introduction to the process for the interested user.

System Compatibility

The two main areas of concern are: interaction between transmitters and receivers related to their operating frequencies, and interactions between transmitters and receivers related to their internal frequencies. The first class of interactions is the more important one and may occur in any group of wireless microphone systems. It is also the one more cumbersome to calculate. The second class of interactions is less problematic and is also relatively easy to predict. However, it is determined by specific system characteristics.

Operating Frequency Interactions: Intermodulation

A single wireless microphone system can theoretically be used on any open operating frequency. When a second system is added it must be on a different operating frequency in order to be used at the same time as the first. This limitation arises from the nature of radio receivers: they cannot properly demodulate more than one signal on the same frequency. In other words, it is not possible for a receiver to "mix" the signals from multiple transmitters. If one signal is substantially stronger than the others it will "capture" the receiver and block out the other signals. If the signals are of comparable strength none of them will be received clearly.

The effect of this is often heard in automobile radios when travelling out of range of one station and into the range of another station at the same frequency. The receiver will switch back and forth between the two stations as their relative signal strength changes, often with considerable noise and distortion. The result is that neither station is listenable when the signals are nearly equal.

If the wireless microphone systems must be on different frequencies, how “different” should they be? The limiting characteristic of the receiver in this regard is its "selectivity" or its ability to differentiate between adjacent frequencies. The greater the selectivity the closer together the operating frequencies can be. The recommended minimum frequency difference ranges from 300kHz to 1.5 MHz, depending on receiver selectivity.

When a third system is added to the group it must be at least the recommended frequency difference from each of the existing systems. However, it is now necessary to consider other potential interactions between the transmitters to insure that all three systems will be compatible with each other. The most important type of interaction is called intermodulation (IM), and it arises when signals are applied to "non-linear" circuits. (See Figure 3-7.)

A characteristic of a non-linear circuit is that its output contains "new" signals in addition to the original signals that were applied to the circuit. These additional signals are called IM products and are produced within the circuit components themselves. The frequencies of IM products are mathematically related to the original transmitter frequencies. Specifically, they consist of sums and differences of the original frequencies, multiples of the original frequencies, and sums and differences of the multiples. Non-linear circuits are intrinsic to the design of wireless components and include the output stages of transmitters and the input stages of receivers. The "mixer" stage at the receiver input is an example of a non-linear circuit: recall that it is designed to produce a "difference" frequency that becomes the intermediate frequency (IF) for subsequent stages.
IM can occur when transmitters are in close proximity to each other. The signal from each transmitter generates IM products in the output stage of the other. These new signals are transmitted along with the original signals and can be picked up by receivers operating at the corresponding IM frequencies. (See Figure 3-10.) IM can also occur when transmitters are operated very close to receivers. In this case IM products are generated in the receiver input stage which can interfere with the desired signal or be detected by the receiver if the desired signal (transmitter) is not present.

The strongest IM products are the two so-called 3rd order products produced by two adjacent transmitters operating at frequency $f_1$ and frequency $f_2$, where $f_1$ is lower than $f_2$. The resulting IM products may be calculated as:

$$IM_1 = (2 \times f_1) - f_2$$
$$IM_2 = (2 \times f_2) - f_1$$

If the interval between $f_1$ and $f_2$ is $F$, then $IM_1 = f_1 - F$ and $IM_2 = f_2 + F$. That is, one IM will appear exactly at interval $F$ above the upper frequency $f_2$ while the other IM will appear exactly at interval $F$ below the lower frequency. For example, if $f_1 = 180$MHz and $f_2 = 190$MHz, then $F = 10$MHz. Thus, IM1 = 170MHz and IM2 = 200MHz. (See Figure 3-8.)

In addition to IM products generated by interaction between two transmitters, other IM products are generated by interaction between three transmitters in a similar fashion. (See Figure 3-9.) In order to avoid potential IM problems most manufacturers recommend a minimum margin of 250 kHz (0.25 MHz) between any 3rd order IM product and any operating frequency. This further restricts available frequency choices as the number of simultaneous systems increases.

It should be apparent from this discussion that the prediction of potential compatibility problems due to IM products is best left to computer programs. The complexity increases exponentially for additional systems: a group of 10 wireless microphone systems involves thousands of calculations. See Appendix A- for details on calculation of IMD products.

**Internal frequency Interactions: LO and IF**

In addition to frequency conflicts due to intermodulation between operating frequencies there are certain other sources of potential conflicts due to the various "internal" frequencies present in the normal operation of transmitters and receivers. These differ from manufacturer to manufacturer and even from system to system from the same manufacturer.

One such source is the local oscillator (LO) of the receiver itself. Although this is a low-level signal which is generally confined within the receiver, it is possible for the local oscillator of one receiver to be picked up by another receiver tuned to that LO frequency if they or their antennas are in close proximity to each other (stacked, for instance). For example, assuming a typical intermediate frequency (IF) of 10.7 MHz, a receiver tuned to 200.7 MHz would have its LO operating at 190.0 MHz. Another receiver tuned to 190 MHz should not be used close to the first receiver because the second unit could pick up the LO of the first, especially if the 190 MHz transmitter is turned off or is operating at a great distance. (See Figure 3-8.) Good design and shielding in receivers and physical separation of receivers will minimize the possibility of LO interference. For multiple units, active antenna splitters will

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**Figure 3-8:** two-transmitter intermodulation
**Figure 3-9:** three-transmitter intermodulation
effectively isolate antenna inputs from each other. However, it is still recommended that operating frequencies be chosen to avoid LO frequencies by at least 250 KHz.

An "image" frequency is another source of possible interference. In a receiver, recall that the frequency of the local oscillator (LO) always differs from the frequency of the received signal by an amount equal to the intermediate frequency (IF). For example, in a low-side injection receiver, the operating frequency is above the local oscillator frequency by an interval equal to the intermediate frequency. When these two frequencies are applied to the mixer section (a non-linear circuit) one of the output frequencies of the mixer is this difference frequency (the IF), which is the tuned frequency of the subsequent IF stage filters.

If the frequency of a second signal is at the same interval below the local oscillator frequency, the difference between this second frequency and the LO frequency would again be equal to the intermediate frequency (IF). The mixer stage does not discriminate between "positive" or "negative" frequency differences. If this second (lower) frequency enters the mixer stage, it will result in another (difference) signal getting to the IF stages and causing possible interference. This lower frequency is called the "image" of the original frequency. For example, assuming an IF of 10.7 MHz, a low-side injection receiver tuned to 200 MHz, would have its LO at 189.3 MHz. A signal from another transmitter at 178.9 MHz would appear as an image frequency since it is 10.7 MHz below the LO frequency (21.4 MHz below the operating frequency).

The image frequency differs from the operating frequency by an amount equal to two times the intermediate frequency (2 x IF). (See Figures 3-12 a & b.) The image frequency will be below the operating frequency for a low-side injection receiver and above the operating frequency for a high-side injection receiver. Thus the image frequency for the typical single conversion receiver is at least 20 MHz away from the operating frequency. Double conversion receivers, which have a relatively high first IF (100 MHz typical), have image frequencies which are even farther (>200 MHz) from the operating frequency. In most cases, the front end of the receiver should be able to reject an image frequency unless it is extremely strong. Nevertheless, it is recommended that operating frequencies be chosen to be at least 250 KHz from any image frequency.

![Figure 3-10: two-transmitter IMD strength vs. transmitter separation](image)

![Figure 3-11: local oscillator (LO) interference](image)
It can be seen that the calculation of both local oscillator (LO) conflicts and image frequency conflicts depends on the specific intermediate frequency (IF) of the receiver. If receivers with different IF values are used together this should (ideally) be taken into account in the compatibility analysis. However, LO leakage is generally negligible in modern receiver design. In addition, the downconversion process, especially with double conversion receivers, places image frequencies well outside the typical front-end filter range. For these reasons, LO conflicts and image frequency conflicts can usually be ignored in the frequency coordination process. Intermodulation (IM) conflicts are the principal calculations that must be considered for successful multi-system compatibility.

**Non-System Radio Interference**

Even though a group of wireless microphone systems may be carefully chosen to avoid mutual interference there always exists the possibility of interference from non-system sources. These sources fall into two categories: broadcast (including television and other defined radio sources) and non-broadcast (narrow band or broadband sources of radio noise). We will look at each of these sources in terms of potential problems and possible solutions.

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**Figure 3-12a:** Image frequency interference (low-side injection)

**Figure 3-12b:** Image frequency interference (high-side injection)
Broadcast Television

In the US, and many other countries, broadcast television is undergoing a transition from analog to digital. This transition affects wireless audio systems in several ways: more "occupied" TV channels, no "open" space in DTV channels, and future "re-allocation" of existing TV channels.

Nearly all full power television stations in the US are now broadcasting only digital signals (DTV). A few stations (mainly along international borders) and some low-power stations still broadcast an analog signal. Though both types of signal occupy similar channel "blocks", the nature of the signal within the channel is quite different. An analog TV transmission consists of three separate signals, each at a specified carrier frequency within a 6 MHz block (in the US). (See Figure 3-13a.) The picture or "video" information is an AM signal at 1.25 MHz above the bottom (low frequency end) of the block. The sound or "audio" information is an FM signal located at 0.25 MHz below the top (high frequency end) of the block. The color or "chroma" information is an AM signal at 3.58 MHz above the video signal. The energy distribution and occupied bandwidth of these three signals is not equal: the video signal has the highest power and widest bandwidth, followed by the audio signal and finally the chroma signal with the lowest power and smallest bandwidth.

A digital TV transmission consists of a continuous signal that occupies the entire 6 MHz block. (See Figure 3-13b.) All of the video, audio, and color information is digitally encoded into this signal along with a variety of other data, control, and secondary audio information. It is possible for the DTV transmission to carry one or more high definition television signal (HDTV) and/or multiple standard definition television signals. The energy distribution within a DTV channel is essentially uniform. However, the average signal level of a DTV transmission is somewhat less than the levels of the video and audio signals in an analog TV transmission.

As indicated previously, the primary users of both high-band VHF and low-band UHF frequencies are broadcast television stations. In the US these are VHF TV channels 7 through 13 and now UHF TV channels 14 through 51. Each TV channel is allotted a 6 MHz block for its transmission. VHF channel 7 begins at 174.0 MHz and extends to 180.0 MHz, channel 8 occupies 180-186 MHz and so on up to channel 13 at 210-216 MHz. UHF channel 14 occupies 470-476 MHz with successive channels up to channel 51 at 692-698 MHz.

The 6 MHz/TV channel block is found in the US, the rest of North America, South America and Japan. Other countries, most of Europe and India for example, use a 7 MHz/TV channel block, while France and China, among others, use an 8 MHz/TV channel block. For analog transmission in these other systems, the video and audio signals are located at the same frequencies relative to the channel boundaries as in the 6 MHz systems, but the frequency of the chroma signal differs slightly in each to accommodate the various color systems: NTSC (6 MHz), PAL (7 MHz), and SECAM (8 MHz). DTV proposals for all systems specify the appropriate TV channel block sizes.

Traditionally, to avoid potential interference between broadcast television stations, regulatory agencies have not...
Selection and Operation of WIRELESS Microphone Systems

allowed adjacent analog TV channel operation in a given geographic area. For example, in the US, if a local analog TV channel 9 existed, then analog channels 8 and 10 would be vacant. These vacant channels could then be used by wireless microphone systems with little concern for television interference.

Historically, the prohibition of adjacent channels guaranteed the existence of certain vacant TV channels in a given area. The advent of DTV has removed this guarantee: DTV channels are allowed to exist adjacent to each other and also adjacent to existing analog TV channels. This has resulted not only in more densely spaced TV channels but also in difficulty using certain "pre-selected" frequency compatibility schemes.

The effects of interference from television broadcast are dependent on the type of television signal (analog or DTV) and the strength of the television signal, as well as the location and the operating frequency of the wireless microphone system. Direct conflicts with any of the three signals that make up an analog TV transmission can produce noise, distortion, and short range or dropout. Pickup of the video or chroma signals (which are AM) may cause distinct "buzz" in the wireless receiver, while pickup of the audio signal (FM) may result in the TV sound being heard.

Direct conflict with a DTV signal also causes short range or dropouts. Though it is not possible for an analog receiver to "hear" anything from a digital transmission, the DTV signal acts as a very strong broadband radio noise source. In some systems this may result in increased noise or distortion in the audio output. Since there are no "unoccupied" frequencies in a DTV signal it is generally not possible to operate anywhere within a very strong DTV channel.

The most effective solution for broadcast television interference is to avoid using frequencies of local active TV channels. Television transmitters may operate at power levels up to one million watts while wireless microphone systems typically have only less than 50 mW (fifty one-thousandths of one watt!) of output power. For this reason it is unwise to choose wireless microphone system frequencies that fall in an active local TV channel block.

"Local is generally considered to be up to 50 or 60 miles, depending on the coverage area of the particular TV transmitter and on the location of the wireless microphone system. Indoor setups are at less risk than outdoor setups because building structures will usually strongly attenuate TV signals. Nevertheless, since the locations and assignments of television stations are well known it is relatively easy to choose wireless microphone system frequencies to avoid them in a particular area.

One consequence of the relatively dense TV channel distribution in the US is that it is not generally possible to use a given set of wireless microphone frequencies everywhere in the country. As the distribution of vacant channels changes from city to city it is almost inevitable that a fixed-frequency system will experience interference from a television station in some location. For this reason, nearly all current wireless systems are "frequency-agile" (tunable) so that their operating frequencies may be changed as needed.

Many manufacturers pre-select groups of wireless microphone system frequencies based on the availability of locally vacant TV channels. If the TV channels are known then it is relatively straightforward to choose appropriate wireless frequencies for that location. Information on TV channel distribution throughout the US is widely available from manufacturers and other sources. It is usually sufficient to indicate the destination of the wireless equipment at the time that it is specified in order to avoid broadcast television interference.

![Figure 3-14: UHF TV band after post-auction transition period ends in July 2020](image-url)
White Space Devices

“White Space Devices” or “WSD’s”, previously called TV Band Devices, are a new type of unlicensed radio system intended primarily to increase broadband internet access in underserved areas. “White Space” refers to TV channels that are not assigned to a TV station in a given geographic area. Of course, most professional wireless audio systems already operate on those same vacant TV channels, raising the possibility of interference from white space systems.

The FCC has defined the operating parameters and regulations for both “fixed” and “portable” White Space Devices. A fixed WSD can operate in a specific location in the VHF or UHF band, on TV channels 2-37. Its power level can vary depending on how congested the spectrum is:

- Up to 40 mW when operating on a single TV channel that is between two channels that are occupied by TV stations
- Up to 100 mW when centered across the boundary of two contiguous vacant TV channels
- Up to 4 watts when operating on a TV channel bordered by one or more vacant TV channels
- Up to 10 watts in “less congested areas” – defined as locations where more than half of the TV channels in the operating band (such as VHF or UHF) are vacant

A portable WSD can operate on TV channels 14-37 only, at the following power levels:

- Up to 40 mW when operating on a TV channel that is adjacent to an occupied channel
- Up to 100 mW on all other channels

Due to their potentially high power, fixed WSD’s have the ability to interfere with wireless microphone operation at a considerable distance. Because they are “always on” and their locations are fixed, however, it should be fairly easy to avoid operating a wireless microphone on the same frequency. Portable WSD’s would obviously be more unpredictable, but would only be a real source of interference within a few hundred feet of a wireless microphone.

As of early 2017 only a small number of fixed devices are known to have been placed in service. With the reduced number of unassigned TV channels in many cities following the incentive auction, it is unclear whether the use of White Space Devices will continue to grow in the near future.

To prevent WSD’s from transmitting on a channel that is assigned to a local TV station, each WSD must have the capability to accurately determine its geographic location and to access an online database of excluded channels for that location. Each time a WSD is activated, it must access the database before transmitting, and again periodically to verify that the channel information is still correct.

In addition to the locations and coverage patterns of local TV stations, the database includes the locations of certain other non-broadcast users of the TV band, such as radio telescope installations, medical facilities where medical telemetry devices are used, and public safety agencies. The database will also list temporary exclusions to protect licensed wireless microphone users on specific frequencies at a specific time and location. High-power WSD’s must observe a 1 kilometer exclusion zone around these users while low-power WSD’s must maintain a 400 meter distance.
Broadcast Radio

High-band VHF wireless FM systems are not generally subject to interference from commercial AM or FM radio stations. Both bands are well below the VHF band and in particular, these systems are not typically sensitive to moderate AM signals. UHF systems are even less likely to respond to commercial radio sources since the UHF band is >300 MHz above the top of the FM band. However, occasional interference, in the form of distortion or short range, can occur in cases of extreme proximity to a high-power commercial radio transmitter.

Other Radio Services

Direct pickup of 2-way radio, pagers, business band, ham-radio etc. is rare. However, since some of these sources can be quite strong locally there is the possibility of interference due to intermodulation or if the source appears as an image frequency. For example, operating a walkie-talkie near a wireless receiver can cause noise, distortion or apparent loss of range.

In particular, the "travelling" wireless microphone system frequencies (169-172 MHz) share space with a number of unpredictable primary users including government (Coast Guard, Secret Service), industry (forestry, hydroelectric), land mobile and paging services. Direct pickup or inter-modulation from these sources is possible in this band. Symptoms may again include noise, loss of range or actual audio pickup.

Some larger US metropolitan areas operate Public Safety radio systems in UHF TV stations 14-20. It is illegal to operate wireless microphone systems on Public Safety channels in those cities. See Appendix B for a detailed list.

Of course nearby use of other (unexpected) wireless microphone systems can create interference through intermodulation or direct frequency conflict. Given the finite number of wireless frequencies available, especially in the "travelling" channels, it is always possible to encounter other wireless microphone users in locations such as hotels, convention centers, sports facilities, and news events.

Remedies for such interference involve identifying the interfering source if possible and relocating the source or the wireless microphone system to reduce proximity. If this is not sufficient it may be necessary to change the operating frequency of the wireless microphone system especially if the interfering source is a primary (licensed) user.

Non-Broadcast Sources

Non-broadcast sources are those that produce radio frequencies only as a by-product of their operation. There are three main types which are of concern to wireless use: digital audio/video devices, digital computers/controllers, and certain AC power equipment. Digital audio devices include:

- Digital video devices include cameras, camcorders, video switchers, video DSPs, and video editing systems.
- Digital computing devices include: microprocessor equipped units (PC’s, calculators), minicomputers (workstations) and main-frame computers. In addition, controllers for lighting, AV presentations, industrial equipment and certain video equipment contain microprocessors.
- The increased use of LEDs in lighting and video displays has become a significant source of local RF interference. Emissions from LED video walls as well as from the associated drive and control equipment can negatively affect range and reliability of all types of wireless audio systems. The radiation from these digital sources is typically wideband noise that decreases gradually at higher frequencies. However, it is sometimes accompanied by stronger peaks at specific frequencies. (See Figure 3-15.)
- Digital devices can produce broad band RFI (radio frequency interference) in close proximity to the source. Any device which carries an FCC type approval label such as "Class B computing device" can be assumed to be a potential source of interference. The audible effect is usually high frequency noise or distortion and it generally only occurs when the receiver is close to the digital device and the transmitter is at a distance. Unfortunately, this is often the case when wireless receivers are located in or near racks of digital gear while the transmitters are being used on a stage.
- The best remedy for this type of interference is to locate the receivers and antennas at least several feet from any digital device. In a rack of different equipment this would suggest mounting the wireless receivers at the top, analog equipment below that and digital equipment at the bottom. In extreme cases, choosing higher wireless frequencies may improve matters. However, as the speed (clock frequency) of digital equipment increases this technique will be less effective.
- Occasional sources of apparent interference are mobile telephones and digital messaging products operating with GSM technology. Although transmit and receive
frequencies of these devices are well outside the bands used by wireless audio equipment they may still cause audible effects. In particular, when GSM devices are in a "handshake" mode or transmitting data, they produce high levels of RFI that can be picked up directly by the audio circuits of wireless microphones as well as wired microphones and other audio devices. This can result in a characteristic buzzing or “chirping” sound in the affected equipment. Although newer professional audio products are more resistant to this effect, the best practice is to turn off any GSM mobile device that may be used within a few feet of any audio device.

Finally, any equipment that uses or controls high voltage or high current AC power can generate radio frequency interference (RFI). Examples include lighting dimmers and some types of gas discharge lamp supplies such as neon or fluorescent ballasts.

Audible effects of this type of source include buzz or hum in the signal. Again, the first remedy is to relocate the offending source or the wireless equipment to minimize pickup. In some cases special filtering may be applied to the various power and connecting cables of both the source and the wireless equipment to block RFI from leaving the source or entering the wireless equipment.

A property of FM reception which can reduce the audibility of many types of interference is the so-called "capture effect." When multiple signals (close to the operating frequency) are present, the strongest signal will capture or lock-in the receiver. If the desired signal is sufficiently strong, the interfering signals may not be heard. Since the strength of the desired signal (the transmitter) is dependent on the operating distance, a nearby transmitter can often overcome weak or distant interference sources.

Range of Wireless Microphone Systems

A logical question concerning wireless performance is the transmission range of various systems. Unfortunately, the answer is much more complicated than a simple distance measurement. Ultimately, the receiver must be able to pick up a "useable" signal from the transmitter. "Useable" means that the strength of the desired signal is within the sensitivity range of the receiver and further that it is sufficiently stronger than (or different from) undesirable signals and RF noise to produce an acceptable signal-to-noise ratio at the audio output of the receiver. Elements that affect useability are the transmitter/antenna, the transmission path, the receiver/antenna and RFI. Some characteristics of these elements are controllable, some are not. (See Figure 3-16.)

Important transmitter characteristics are power output and antenna efficiency. Maximum power is limited by government regulations and battery capability. Antenna efficiency is limited by size and design. Recall that the efficiency of typical wireless transmitter antennas is fairly low, about 10% or less for VHF. This means that for a 50 mW VHF transmitter the effective radiated power (ERP) is less than 5 mW. This may be further attenuated by proximity to the body or other lossy objects.

Important receiver characteristics are antenna efficiency, receiver sensitivity and the ability of the receiver to reject unwanted signals and noise. Antenna efficiency is again limited by size and design but receiver antennas tend to be much more efficient than transmitter antennas since they can be made large enough to be better tuned to the proper frequency. Other receiver characteristics are limited by design. Both elements are limited by cost.

The transmission path is characterized by distance, intervening obstructions and propagation effects. Losses due to these characteristics are generally frequency dependent: the higher the frequency the greater the loss. Once the operating frequency is chosen, only the path length and antenna locations are controllable. These are usually limited by the application itself. Under good conditions (line-of-sight) at a distance of about 100 ft. the field strength of the signal from a 50 mW transmitter is on the order of 1000 uV/m, well within the range of sensitivity of a typical receiver.

Finally, RFI is characterized by its spectrum, that is, its distribution of amplitude and frequency. It typically consists of both broadband noise and discrete frequencies. However, its strength can be comparable to or greater than the desired signal in poor conditions. Except for a few predictable sources it is largely uncontrollable.

Rather than quote a specific maximum operating distance most manufacturers of wireless microphone systems give a "typical" range. For systems of the type discussed here (10-50 mW, VHF or UHF) the typical range may vary from 100 ft. to 1000 ft. The lower number represents a moderately severe environment while the upper figure might be achieved in absolute ideal conditions. Extremely poor conditions could result in a range of only 50 feet or less. It is impossible to accurately predict the range of an arbitrary wireless microphone system in an arbitrary application.
CHAPTER FOUR

WIRELESS SYSTEM SELECTION AND SETUP

System Selection

The proper selection of a wireless microphone system consists of several steps based on the intended application and on the capabilities and limitations of the equipment required for that application. It should be remembered that while wireless microphone systems cannot ultimately be as consistent and reliable as wired systems, the performance of currently available wireless can be very good, allowing excellent results to be obtained. Following these steps will insure selection of the best system(s) for a given application.

1) Define the application. This definition should include the intended sound source (voice, instrument, etc.) and the intended sound destination (sound system, recording or broadcast). It must also include a description of the physical setting (architectural and acoustic features). Any special requirements or limitations should also be noted: cosmetics, range, maintenance, other possible sources of RF interference, etc. Finally, the desired performance level must be defined: radio quality, audio quality, and overall reliability.

2) Choose the microphone (or other source) type. The application will usually determine which microphone physical design is required: a lavaliere or clip-on type attached to clothing, or a head-worn type, both for hands-free use; a handheld type for a vocalist or when the microphone must be passed around to different users; a connecting cable when an electric musical instrument or other non-microphone source is used. Other microphone characteristics (transducer type, frequency response, and directionality) are dictated by acoustic concerns. As mentioned earlier, the microphone choice for a wireless application should be made using the same criteria as for a wired application.

3) Choose the transmitter type. The microphone choice will usually determine the required transmitter type (hand-held, bodypack or plug-on), again based on the application. General features to consider include: antenna style (internal or external), control functions and location (power, muting, gain, tuning), indicators (power, battery condition), batteries (operating life, type, accessibility), and physical description (size, shape, weight, finish, material). For handheld and plug-on types interchangeability of microphone elements may be an option. For bodypack transmitters, inputs may be hard-wired or detachable. Multi-use inputs are often desirable and may be characterized by connector type, wiring scheme and electrical capability (impedance, level, bias voltage, etc.).

4) Choose the receiver type. The basic choice is between diversity and non-diversity. For reasons mentioned in the receiver section above, diversity receivers are recommended for all but the most budget-conscious applications. Though non-diversity types will work well in many situations, the insurance provided by the diversity receiver against multipath problems is usually well worth the somewhat higher cost. Other receiver features that should be considered are: controls (power, output level, squelch, tuning), indicators (power, RF level, audio level, frequency), antennas (type, connectors), electrical outputs (connectors, impedance, line/microphone/headphone level, balanced/unbalanced). In some applications battery power may be required.

5) Determine the total number of systems to be used simultaneously. This should take into account future additions to the system: choosing a system type that can only accommodate a few frequencies may prove to be an eventual limitation. Of course, the total number should include any existing wireless microphone systems with which the new equipment must work.

6) Specify the geographic location in which these systems will be used. This information is necessary in the next step to avoid possible conflict with broadcast television frequencies. In the case of touring applications, this may include cities inside and outside of the US.

7) Coordinate frequencies for system compatibility and avoidance of known non-system sources. Consult the manufacturer or a knowledgeable professional about frequency selection and integration of the planned number of systems. This should be done even for single systems and must certainly be done for any multiple system installation to avoid potential interference problems. Frequency coordination includes the choice of operating band (VHF and/or UHF) and choice of the individual operating frequencies (for compatibility and avoidance of other transmissions). For fixed locations choose frequencies in unused TV channels. For touring applications, it may be necessary to carry additional systems on alternate frequencies, though this is only practical for a small number of channels. The preferred approach for touring is to use frequency-agile (tunable) units to insure the required number of systems at all venues.

8) Specify accessory equipment as needed. This may include remote antennas (1/2 wave, 5/8 wave, directional), mounting hardware (brackets, ground-planes), antenna splitters (passive, active), and antenna cables (portable, fixed). These choices are dependent on operating frequencies and the individual application.
System Setup: Transmitter

Transmitter setup first requires optimizing the source-to-transmitter interface. Sources include dynamic and condenser microphones, electronic musical instruments and general audio sources such as mixer outputs, playback devices, etc. The output signal of each of these sources is characterized by its level, impedance and configuration (balanced or unbalanced). For sources such as condenser microphones, some type of power (phantom or bias) may be required.

The transmitter may be a bodypack, plug-on or handheld type and its input will also have a characteristic level, impedance and configuration (balanced or unbalanced). It may be capable of supplying power to the source. The interface can consist of some type of connector or it may be hard-wired, either internally or externally. (See Figure 4-1.)

The simplest interface is the handheld transmitter. This design should insure that the microphone element is already optimally integrated (electrically and mechanically) with the transmitter. The only choice involves systems that offer a selection of microphone elements. If each is equipped for proper interface the decision should be made based on the performance characteristics of the microphone element for the intended application.

The plug-on transmitter offers a range of interface possibilities. Mechanically, the 3-pin XLR type connector is standard but the electrical characteristics of the chosen microphone and transmitter combination must be considered. The input impedance of the transmitter should be higher than the microphone output impedance. All transmitters of this type will work with typical low-impedance dynamic microphones. If the transmitter input impedance is high enough (>10,000 ohms) a high impedance microphone may also be used. Most plug-on transmitters will work with either balanced or unbalanced microphone outputs.

Some plug-on transmitters are also capable of supplying “phantom power” to a condenser microphone. This is only possible with a balanced transmitter input and a balanced microphone output. Even then, the transmitter must supply at least the minimum phantom voltage required by the microphone (usually between 11 and 52 volts DC). If less than the minimum is available, the condenser microphone performance may be compromised with less headroom or more distortion. This is not a concern with dynamic microphones (which do not require power) or with condenser microphones powered by an internal battery.

The bodypack transmitter presents the greatest range of possible interfaces. The simplest arrangement is the hard-wired lavalier or headset microphone. Again, it can usually be assumed that this design already provides the optimum interface for the components provided. If various hardwired microphone choices are offered, the selection should be based on the intended application.

Most bodypack transmitters are equipped with an input connector to allow the use of a variety of microphones and other input sources. (See Figure 4-2.) Microphones and input cables supplied by a manufacturer with a given wireless microphone system can be assumed to be compatible with that system. However, they may not be directly compatible with wireless microphone systems from other manufacturers. At a minimum, a connector change is often required. In many cases, additional circuitry or modifications to components will be necessary. A few combinations simply will not work.

In order to determine the suitability of a particular microphone for use with a particular transmitter it is first necessary to determine the connector type(s) involved. Connectors include eighth-inch and quarter-inch phone jacks as well as a variety of multi-pin designs.
Next, the wiring of the microphone connector and the wiring of the transmitter connector must be compared. Unfortunately, there is no standard input connector, and further, the wiring scheme of the same connector may differ from one manufacturer to another. A quarter-inch input jack is usually wired unbalanced with the audio signal at the tip and shield on the sleeve. The typical multi-pin input on a body-pack transmitter has at least one pin for the audio signal and one pin for shield or ground. There may be other pins to provide "bias" (a DC voltage for a condenser microphone element) or to provide an alternate input impedance. Some transmitters have additional pins to accept audio signals at different levels or to provide a combination audio + bias for certain condenser elements.

The electrical characteristics of the microphone and transmitter should then be compared: the output level of the microphone should be within the acceptable input level range of the transmitter and the output impedance of the microphone should be less than the input impedance of the transmitter. In addition, the input configuration of most bodypack units is unbalanced. Microphones intended for use with wireless are also invariably unbalanced, though a balanced output dynamic microphone can usually be accommodated with an adapter cable.

If the microphone has a condenser element and does not have its own power source then the transmitter must supply the required bias voltage. Most transmitters provide about 5 VDC, suitable for a typical electret condenser element, though some elements may require as much as 9 VDC. In this case, it is sometimes possible to modify the transmitter to provide the higher voltage.

Many condenser elements and associated transmitters use a two-conductor-plus-shield hookup in which the audio is carried on one conductor and the bias voltage on the other. A few condenser elements and some transmitters use a single-conductor-plus-shield arrangement in which the audio and bias voltage are carried on the same conductor. Interfacing a microphone of one scheme with a transmitter of the other may require modification of one or both components.

In general, for non-standard combinations, it is best to directly contact the manufacturer of the wireless microphone system and/or the manufacturer of the microphone to determine the compatibility of the desired components. They can provide the relevant specifications and can usually describe any limitations or necessary modifications.

Musical instruments such as electric guitars and basses can have output levels from a few millivolts (microphone level) for instruments with passive pickups to a few volts (line level) for those with active pickups. The transmitter must be capable of handling this dynamic range to avoid overmodulation or distortion.

Ordinary (passive) magnetic instrument pickups have a high output impedance and require a transmitter input impedance of about 1 Megohm to insure proper frequency response. Active (powered) pickups have fairly low output impedance and will work with almost any transmitter input impedance of 20,000 ohms or greater. Piezoelectric pickups have very high output impedance and require a 1-5 Megohm transmitter input impedance to avoid loss of low frequencies.

Mixers and playback devices produce line level outputs. These sources typically have low-to-medium output impedance and may be balanced or unbalanced. They can sometimes be interfaced with a simple adapter cable. However, these high level input sources often require additional (external or internal) attenuation to prevent overload of the transmitter input, which is usually expecting a mic-level signal.

Once the source/transmitter interface has been optimized, audio gain adjustment should be performed. The gain control on most transmitters may be a small potentiometer or a menu-driven adjustment. There may also be a switchable attenuator that can be engaged to accommodate very high-level signals. Some digital systems do not have adjustable gain controls on the transmitter. For those systems, gain adjustment is done at the receiver only.
The control(s) should be adjusted so that the loudest sound level (or highest instrument level) in actual use produces full modulation of the radio signal. This is usually determined by speaking or singing into the microphone (or playing the instrument) while observing audio level indicators on the receiver. Typically, an audio peak LED will indicate full (or nearly full) modulation. A few designs have peak indicators on the transmitters themselves. In systems that indicate peaks at less than full modulation, this LED may light fairly often. For systems that indicate full modulation, this should light only briefly at maximum input levels. In either case, sustained peak indication requires reducing input sensitivity or level to avoid audible distortion.

If the transmitter is equipped with a compander system (noise reduction) defeat switch make sure that it is set to the same mode as the receiver. The only situation in which this system would be defeated is with the use of a receiver that is not equipped with compander circuitry. Non-companded analog wireless systems are used primarily for audio measurement purposes.

For tunable transmitters, make sure that the transmitter is set to the same frequency as the receiver.

The last step in transmitter setup is placement. Placement of a handheld or plug-on system is essentially the same as for a wired microphone of the same type. The unit may be mounted on a stand, boom or fishpole with an appropriate stand adapter, or it may be handheld.

Bodypack transmitter placement is dependent on the particular application. If the input source is a microphone, such as a lavaliere or headset, the bodypack is normally clipped to a belt or pants waistband. It may be attached in other ways as long as the antenna is allowed to extend freely. Insure that there is adequate access to the controls if necessary and that the connecting cable, if any, has enough length to permit the source and the transmitter to be located as desired. When the input is a musical instrument, it is often possible to attach the transmitter directly to the instrument or to its strap as in the case of an electric guitar.

For all types of transmitters, insure that the antenna is securely attached and positioned for maximum efficiency. Wire antennas should be fully extended. The hand should not cover external antennas on handheld transmitters. (See Figure 4-3.)

As much as possible, proper transmitter placement should avoid large metal objects and previously mentioned sources of RF such as digital devices, other wireless transmitters and mobile telephones. If an individual is using more than one wireless system at the same time, such as a wireless head-set and a wireless musical instrument, or is wearing a wireless personal monitor receiver, the devices should be kept as far apart as practical to minimize interaction.

System Setup: Receivers

Receiver setup involves two interfaces: antenna-to-receiver and receiver-to-sound system. (See Figure 4-4.)

Audio Interface

Here we will discuss the sound system interface. Remember that the basic function of a wireless microphone system is to replace the connecting cable between the source and the sound system. In the typical case, the output of the wireless receiver will resemble the output of the original source both electrically and physically. That is, most wireless microphone receivers have a balanced, low impedance, microphone level output, usually on a standard 3-pin XLR-type audio connector. This can be connected to a standard microphone input of a sound system using an ordinary balanced microphone cable. (See Figure 4-5.)
Some receivers, particularly those designed for use with electric instruments, may be equipped with a quarter-inch phone jack instead of (or in addition to) an XLR connector. Normally, this output is an unbalanced, low or medium impedance, microphone or instrument level signal. It can be directly connected to the input of an instrument amplifier using a standard shielded instrument cable.

In addition, a few receivers have line level outputs available. These can be connected to line or aux level inputs of sound systems equipped with similar types of input connectors.

If it is desired (or necessary) to connect one type of output to a different type of input a few possibilities should be considered. For a balanced XLR output to an unbalanced quarter-inch input an adapter which connects Pin 2 of the XLR to the tip of the phone plug and connects Pin 1 and Pin 3 of the XLR to the sleeve of the phone plug may be used. A similar adapter (with appropriate XLR connector) may be used to connect an unbalanced quarter-inch output to a balanced XLR input. Simple adapters of this type will usually work if the levels and impedances of the outputs and inputs are compatible.

In some cases simple adapters cannot be used due to significant impedance or level differences. In addition, the quarter-inch phone-to-XLR hookups just described (which cause the circuit to be unbalanced) can occasionally create audible hum problems due to ground loops between the receiver and the sound system. In either case, the use of a transformer may offer a solution. It can provide the proper transition between different impedances and between balanced and unbalanced circuits. The transformer also allows ground loops to be eliminated by lifting the shield connection at the source end of the balanced cable.

Finally, the presence of phantom power at the balanced microphone input of the sound system must be considered. If the receiver output is unbalanced, phantom power may cause noise or distortion in the signal. Phantom power should be turned off at that input if possible. If not, a suitable transformer or an adapter with capacitors will block the voltage in the connecting path. However, if the receiver output is balanced, phantom power is usually not a problem though a manufacturer may specify the maximum voltage that the receiver can tolerate. A few receivers present a significant load to the phantom source. This can result in a lowering of the phantom voltage at other inputs on a mixer that has insufficient isolation of the phantom supply between inputs.

Many digital and some hybrid receivers may also be equipped with digital signal outputs. These may include AES3/AES-EBU signals on XLR connectors and/or various digital audio network signals on Ethernet connectors. Two AES3/AES-EBU signals can be carried on a single balanced two-conductor cable in a point-to-point connection. Up to 256 digital audio network signals can be carried on Cat5e/Cat6 Ethernet cables across standard Ethernet networks, depending on system protocols. Details of connection and routing requirements are specific to those protocols and must be followed properly for successful operation.

Once the receiver has been properly connected, then the sound system controls may be set. The first control adjustment on a receiver is the output level. This usually consists of a rotary pot and possibly a switch to select microphone or line level. The general procedure is to set the output level so that it is approximately the same as that of a wired source of the same type. This will provide normal gain structure in the rest of the sound system. Though microphone level is most common, line level can be appropriate for long cable runs or for driving line level devices such as equalizers, crossovers or power amplifiers. On most receivers, the audio level indicators are pre-volume control and are unaffected by receiver volume control settings. Use the indicators on subsequent equipment to gauge the actual output level.

Squelch Adjustment

Another receiver adjustment is the squelch control. Recall from the previous discussion that the function of the squelch circuit is to mute the audio output of the receiver when the transmitted signal is lost or becomes unacceptably noisy. Depending on the type of squelch system used (threshold squelch, noise squelch, tone-key squelch) the adjustment procedure will vary:

If a simple threshold squelch is used, adjustment may be required if the radio background noise level changes substantially. This would be indicated by loud “white noise” from the receiver output when the transmitter is turned off or drops out.

If noise squelch is used, no adjustment is normally necessary. Noise squelch mutes the receiver based on the signal-to-noise quality of the audio signal. The receiver will generally not produce noise in the absence of the transmitter signal. Setting the squelch above the default position will force the receiver to mute for mildly noisy signals, which will reduce the effective range somewhat. Setting the squelch below the default position will allow a noisier signal to be received, which may increase the effective range.

A receiver equipped with a tone-key squelch system also does not normally require adjustment. The receiver will only respond to a signal that contains the appropriate pilot “tone.” The squelch may be varied from its default position with the same results as for the noise squelch system above.
Other receiver controls may include monitor (headphone) level, indicator selectors, channel selectors, etc. These may be set as desired for a particular application. If there is a compander (noise reduction) defeat switch, make sure that it is set to the same mode as the transmitter. Again, there is no reason to defeat the compander in the receiver unless the transmitter is not equipped with compander circuitry. If the receiver is tunable, make sure that it is set to the same frequency as the transmitter. Some receivers are now capable of automatically scanning for a clear channel.

Receiver Mounting and Placement

Proper placement of receivers involves both mechanical and electrical considerations. Mechanically, wireless receivers are usually designed to be used like other standard rackmount products. The electrical concerns are possible RF interference and possible hum or other electrical noise induced in the audio circuits. Receivers should be kept away from RF noise sources such as digital processors, computers and video equipment. They should also be separated from large AC sources such as power supplies for high current or high voltage equipment as well as lighting dimmers, fluorescent light ballasts and motors.

If wireless receivers are mounted in racks with other equipment it is best to place them with low-power analog devices nearby and potentially troublesome devices farther away or in a separate rack. In particular, if other wireless transmitting devices such as personal monitor transmitters or wireless intercom transmitters are used, it is strongly recommended that they be mounted in a different rack. Antennas from these transmitters should also be at a sufficient distance from receiver antennas. Obviously, if receivers are placed in metal racks or mounted between other metal devices it will be necessary to make sure that antenna function is not compromised.

System Setup: Receiver Antennas

Setup of receiver antennas involves first the antenna-to-receiver interface and then antenna placement. The simplest case is a receiver with the antenna(s) permanently attached. The antenna is typically a quarter-wave telescoping or possibly “rubber ducky” type. Receivers with non-detachable antennas should be placed on an open surface or shelf, in line-of-sight to the transmitter, for proper operation. They are often not suitable for rack mounting except perhaps as a single unit at the top of a rack and then only if the antennas are mounted on the front of the receiver or can project through the top of the rack.

A receiver with detachable antennas offers more versatility in setup. In most cases the antennas attach to the rear of the receiver. If the receiver is to be mounted in a metal rack the antennas must be brought to the outside of the rack. Some designs allow the antennas to be moved to the front of the receiver, while others provide an accessory panel for antenna relocation. Again, the receiver should be mounted high enough in the rack so that the antennas are essentially in the open.

Here are some general rules concerning setup and use of receiver antennas:

1) **Maintain line-of-sight between the transmitter and receiver antennas as much as possible, particularly for UHF systems.** Avoid metal objects, walls, and large numbers of people between the receiving antenna and its associated transmitter. Ideally, this means that receiving antennas should be in the same room as the transmitters and elevated above the audience or other obstructions. (See Figure 4-6.)

2) **Locate the receiver antenna so that it is at a reasonable distance from the transmitter.** A minimum distance of about 5 meters is recommended to avoid potential intermodulation products in the receiver. The maximum distance is not constant but is limited by transmitter power, intervening objects, interference, and receiver sensitivity. Ideally, it is better to have the antenna/receiver combination closer to the transmitter (and run a long audio cable) than to run a long antenna cable or to transmit over excessively long distances.

3) **Use the proper type of receiver antenna.** A quarter-wave antenna can be used if it is mounted directly to the receiver, to an antenna distribution device or to another panel, which acts as a ground-plane. If the antenna is to be located at a distance from the receiver, a half-wave antenna is recommended. This type has somewhat increased sensitivity over the quarter-wave and does not require a ground-plane. For installations requiring more distant antenna placement or in cases of strong interfering sources it may be necessary to use a directional (Yagi or log-periodic) antenna suitably aimed. Telescoping antennas should be extended to their proper length.
4) Select the correctly tuned receiver antenna(s). Most antennas have a finite bandwidth making them suitable for receivers operating only within a certain frequency band. When antenna distribution systems are used, receivers should be grouped with antennas of the appropriate frequency band as much as possible. For the VHF range: if the receiver frequencies span two adjacent antenna bands, the longer (lower frequency) antennas should be used. If the range spans all three antenna bands, one long antenna and one short antenna should be used (no middle length antenna). For the UHF range: receivers should only be used with antennas of a matching range.

5) Locate diversity receiver antennas a suitable distance apart. For diversity reception the minimum separation for significant benefit is one-quarter wavelength (about 30 cm. for VHF and about 10 cm. for UHF). The effect improves somewhat up to a separation of about one wavelength. Diversity performance does not change substantially beyond this separation distance. However, in some large area applications, overall coverage may be improved by further separation. In these cases one or both antennas may be located to provide a shorter average distance to the transmitter(s) throughout the operating area.

6) Locate receiver antennas away from any suspected sources of interference. These include other receiver and transmitter antennas as well as sources mentioned earlier such as digital equipment, AC power equipment, etc.

7) Mount receiver antennas away from metal objects. Ideally, antennas should be in the open or else perpendicular to metal structures such as racks, grids, metal studs, etc. They should be at least one-quarter wavelength from any parallel metal structure. All antennas in a multiple system setup should be at least one-quarter wavelength apart.

8) Orient receiver antennas properly. A non-diversity receiver should generally have its antenna vertical. A diversity receiver can benefit from antennas oriented about 90 degrees apart. Yagi and log-periodic types should generally be oriented with their elements vertical.

9) Use the proper antenna cable for remotely locating receiver antennas. A minimum length of the appropriate low-loss cable equipped with suitable connectors will give the best results. Refer to the chart presented earlier. Because of increasing losses at higher frequencies, UHF systems may require special cables.

10) Use an antenna distribution system when possible. This will minimize the overall number of antennas and may reduce interference problems with multiple receivers. For two receivers a passive splitter may be used. For three or more receivers active splitters are strongly recommended. Verify proper antenna tuning as mentioned above. Antenna amplifiers may be required with long cable runs.

System Setup: Batteries

Always use fresh batteries of the correct type in the transmitter and/or receiver. Most manufacturers recommend only alkaline type batteries for proper operation. Alkaline batteries have a much higher power capacity, more favorable discharge rate and longer storage life than other types of single-use batteries such as carbon-zinc. Alkaline types will operate up to 10 times longer than so-called “heavy duty” non-alkaline cells. They are also far less likely to cause corrosion problems if left in the unit. Consider bulk purchase of alkaline batteries to get the greatest economy: they have a shelf life of at least one year.

The battery condition should be determined before system use and checked periodically during use, if possible. Most transmitters are equipped with a battery status indicator of some kind that will at least indicate a go/no-go or some minimum operating time. Some units have a “fuel gauge” that can allow more precise indication of remaining battery life. A few models even have the capability of transmitting battery condition information to the receiver for remote monitoring.

Rechargeable batteries may be used in wireless microphones with some reservations. These reservations are dependent on the battery size and on the actual chemistry of the battery. The conventional rechargeable battery uses a Ni-Cad (nickel-cadmium) cell or Ni-Mh (nickel-metal-hydride) cell. The voltage of an individual Ni-Cad or Ni-Mh cell is 1.2 volts rather than the 1.5 volts of an alkaline cell. This is a 20% lower starting voltage per cell. For systems using AA or AAA size batteries, this lower starting voltage may not be an issue because most transmitters using these battery sizes have internal voltage regulators that can compensate. High capacity Ni-Mh single cell (AA or AAA) batteries are available with operating times that are comparable to single cell alkaline types.

However, the standard alkaline 9-volt battery is made up of six cells in series, which yields an initial voltage of at least 9 volts. Typical continuous operating time for a 9-volt alkaline battery in a wireless microphone is about eight hours. The least expensive “9-volt size” rechargeable also has six cells, giving it an initial voltage of only 7.2 volts. When combined with its lower power capacity the operating time may be less than 1/20 of an alkaline, only about 15 minutes in some units. The “better” 9-volt size rechargeable has seven cells (8.4 volts initial), but still has significantly less power capacity than an alkaline. Operating time for these types may be as little as two hours compared to eight hours for an alkaline 9-volt battery (See Figure 4-7).

It is possible to obtain high performance 9-volt size Ni-Mh batteries that approach the power capacity of an alkaline. These may offer up to six hours of operation.
A battery chemistry that exceeds alkaline capacity is the lithium-ion (Li-on) or lithium-polymer (Li-polymer) type. Not only do these batteries offer higher power density than alkalines, they offer better charging characteristics and longer service life than Ni-Cad or Ni-Mh types. In addition, custom versions of the Li-on contain an integrated circuit that monitors both the short term and long term state of the battery. This allows extremely accurate battery voltage measurement as well as maximum battery life. The Li-on battery is becoming the standard rechargeable chemistry for many professional applications.

If it is decided to use rechargeable batteries, battery management is very important. For systems in daily service a minimum of two batteries per unit is recommended due to the charging time: one charging, and one in use. In addition, Ni-Cad batteries must periodically be completely cycled to get maximum service life and avoid developing a short discharge “memory effect.” Generally, Ni-Mh and Li-on types do not exhibit memory effect. However, for maximum performance from any rechargeable battery it is necessary to use a high-quality charger that is designed for the specific battery type. Improper charging can impair or even damage rechargeable batteries prematurely.

Ultimately, the long-term potential savings in battery cost must be weighed against the expected operating time, initial investment and ongoing maintenance requirements for rechargeable batteries.

### Pre-Show System Checkout and Show Operation

Good practice with any wireless microphone system involves a checkout of the system ahead of use as well as monitoring the system during use. To insure a representative checkout, all other production equipment should be operating in normal mode, generating a true “show” environment. This may reveal potential problems that are not apparent in a wireless-system-only test.

The Pre-Show Checkout procedures listed on pg. 48 are intended to insure proper operation for each individual wireless system as well as for all systems simultaneously. Both RF performance and audio performance should be evaluated. This will demonstrate the proper function of the antenna system, the accuracy of the frequency coordination, the optimum audio gain structure, as well as the battery condition.

The Show Operation procedures are intended to monitor and maintain proper function during use. This should include monitoring RF levels, audio levels, and battery levels, generally via indicators on receivers. Equipment adjustments can be made as needed. During the show, un-needed transmitters should be turned off. It is also strongly suggested that unused receivers (receivers without active transmitters) should be muted so that noise from unexpected interference sources is not heard.

### Networkable Wireless Systems

It was noted in the digital wireless section that digital control circuits are now common in wireless systems, particularly frequency-agile designs. Various transmitter and receiver functions are handled by embedded micro-processors. Display information is also digitally generated. One result of these internal technologies is the possibility of external monitoring and control of wireless systems.
Selection and Operation of Wireless Microphone Systems

Although this goal could be achieved with proprietary controls and displays, the most common wireless system interface device is the personal computer. By using such a universal platform it is possible to monitor and control large-scale wireless installations from both local and remote points. Hookup is usually through either USB or Ethernet connections.

Typical monitor capabilities include battery condition, RF signal strength, audio level and diversity status. Typical receiver control functions include frequency selection, squelch level set, and alphanumeric channel naming. In addition, some systems can automate the selection of appropriate operating frequencies without prior knowledge of the RF environment.

Such systems have a set-up mode during which the receiver functions as an automatic radio scanner to find open channels. This scan-mode can be under the control of an onboard microprocessor in an individual receiver or externally controlled by a computer. The receiver can also be made to function as a simple RF spectrum analyzer for interference problems or as an RF level recorder for uncovering possible antenna problems in a pre-show walk-through.

Finally, computer data management capability permits creating and recalling “scenes” or setups for touring, rentals, and other temporary applications, as well as logging of wireless system performance. All of this information can be reviewed as needed for troubleshooting or maintenance. Presently, computer control can assist in setup and operation of even moderate-sized wireless rigs and is the only practical way of integrating very large wireless installations.

"Intelligent" Wireless Systems

An extension of wireless system control is the ability to communicate with transmitters through a secondary radio link. These “back-channel” techniques typically operate in the 2.4 GHz range in a manner that is somewhat similar to Wi-Fi as a two-way data link. This gives the user the ability to control the transmitter remotely. Audio gain, transmitter power, and even transmit frequency can be adjusted at a distance either from an associated receiver or other outboard control device such as a computer.

In addition, receivers have been developed that can recognize an interfering signal. When combined with remote control of transmitters, it is now possible for such a receiver to automatically retune itself and the associated transmitter to a clear frequency with almost no interruption of the audio signal.

When combined with appropriate equipment, such a system can find and set all channels to a compatible frequency set and then continue to monitor and change frequencies as needed during an event to maintain continuous operation even in very dynamic RF environments.

Pre-Show Checkout:

1) Verify good batteries in all transmitters.
2) Turn on all receivers (without transmitters) and all antenna distribution equipment. All receivers should show little or no RF activity.
3) Turn on individual transmitters one at a time to verify activation of associated receiver. Transmitters should all be at a comparable distance (at least 5 meters) from receiving antennas. Off-channel receivers should show little or no RF activity.
4) Turn on all transmitters (with receivers) to verify activation of all receivers. Transmitters should all be at a comparable distance (at least 5 meters) from receiving antennas and at least 1 meter from each other.
5) Turn off transmitters one at a time to verify that associated receiver indicates little or no RF activity.
6) Perform a stationary listening test with each individual system one at a time to verify proper audio level settings.
7) Perform a listening test around the performance area with each individual system one at a time to verify no dropouts.
8) Perform a listening test around the performance area with each individual system while all systems are on to verify no audible interference or dropouts.

Show Operation:

1) Monitor battery levels in all transmitters.
2) Monitor RF levels and audio levels for all receivers.
3) Do not activate unneeded transmitters or their associated receivers.
4) Active transmitters should usually be left on throughout the event. Use audio mute on transmitter if necessary.
5) If an active transmitter must be turned off, mute the associated receiver.
6) Mute the sound system audio channel for any receiver without an active transmitter.
7) Maintain line-of-sight from transmitter antennas to receiver antennas.
8) Maintain transmitter-to-receiver antenna distance of at least 5 meters.
9) Maintain transmitter-to-transmitter distance of at least 1 meter if possible.
10) Operate transmitters in the same general performance area.
Troubleshooting Wireless Microphone Systems

Even when wireless microphone systems appear to be properly selected and set up, problems may arise in actual use. While it is not practical here to offer comprehensive solutions for all possible situations, some general guidelines are suggested.

Though problems with wireless microphone systems eventually show up as audible effects, these effects can be symptoms of audio and/or radio problems. The object of troubleshooting in either situation is first to identify the source of the problem and second to reduce or eliminate the problem.

The following abbreviations are used in these charts:
AF-audio frequency, RF-radio frequency, RFI-radio frequency interference, TX-transmitter, RCV-receiver

A common symptom in multiple system operation is apparent activation of two receivers by a single transmitter. This can be due to one of several causes: operating frequencies the same or too close, crystal harmonics, transmitter at the image frequency of the second receiver, IM with an unknown source, etc. If activating the second transmitter results in proper operation of both systems, this effect can usually be ignored. Recommended operating procedure is to turn up a receiver only when its transmitter is active. If it is desired to allow open receivers without transmitters, readjusting the squelch settings may suffice. Otherwise, the operating frequencies may have to be changed.

### Troubleshooting Guide

<table>
<thead>
<tr>
<th>Conditions: TX on, RCV on, single system</th>
<th>Symptom</th>
<th>TX - RCV Distance</th>
<th>Possible cause</th>
<th>Action</th>
</tr>
</thead>
<tbody>
<tr>
<td>No AF signal and no RF signal</td>
<td>any</td>
<td>low TX battery voltage</td>
<td>replace battery</td>
<td></td>
</tr>
<tr>
<td>No AF signal and no RF signal</td>
<td>any</td>
<td>TX and RCV tuned to different frequencies</td>
<td>retune one or both units</td>
<td></td>
</tr>
<tr>
<td>No AF signal and no RF signal</td>
<td>average</td>
<td>multipath dropout</td>
<td>use diversity RCV or reposition TX and/or RCV</td>
<td></td>
</tr>
<tr>
<td>No AF signal and no RF signal</td>
<td>long</td>
<td>out of range</td>
<td>move TX closer to RCV</td>
<td></td>
</tr>
<tr>
<td>No AF signal but normal RF signal</td>
<td>any</td>
<td>TX muted</td>
<td>unmute TX</td>
<td></td>
</tr>
<tr>
<td>No AF signal but normal RF signal</td>
<td>any</td>
<td>microphone or other input source</td>
<td>check input source</td>
<td></td>
</tr>
<tr>
<td>Distortion with no AF peak indication</td>
<td>any</td>
<td>low TX battery voltage</td>
<td>replace battery</td>
<td></td>
</tr>
<tr>
<td>Distortion with AF peak indication</td>
<td>any</td>
<td>excessive TX input level</td>
<td>decrease source level or TX input level</td>
<td></td>
</tr>
<tr>
<td>Distortion with AF peak indication in subsequent equipment</td>
<td>any</td>
<td>excessive RCV output level</td>
<td>decrease RCV output level</td>
<td></td>
</tr>
<tr>
<td>Noise with low AF signal and normal RF signal</td>
<td>any</td>
<td>insufficient TX input level</td>
<td>increase source level or TX input level</td>
<td></td>
</tr>
<tr>
<td>Noise with low AF signal and normal RF signal</td>
<td>any</td>
<td>strong RFI</td>
<td>identify source and eliminate, or change frequency of wireless microphone system</td>
<td></td>
</tr>
<tr>
<td>Noise with normal AF signal and low RF signal</td>
<td>average</td>
<td>moderate RFI</td>
<td>increase squelch setting until RCV mutes</td>
<td></td>
</tr>
<tr>
<td>Noise with normal AF and RF signals</td>
<td>any</td>
<td>very strong RFI</td>
<td>identify source and eliminate, or change frequency of wireless microphone system</td>
<td></td>
</tr>
<tr>
<td>Intermittent AF signal and low RF signal</td>
<td>long</td>
<td>out of range</td>
<td>move TX closer to RCV</td>
<td></td>
</tr>
<tr>
<td>Intermittent AF signal and low RF signal</td>
<td>long</td>
<td>insufficient antenna gain</td>
<td>use higher gain antenna</td>
<td></td>
</tr>
<tr>
<td>Intermittent AF signal and low RF signal</td>
<td>long</td>
<td>excessive antenna cable loss</td>
<td>use low loss cable and/or less cable</td>
<td></td>
</tr>
<tr>
<td>Intermittent AF and RF signals</td>
<td>average</td>
<td>multipath interference</td>
<td>use diversity RCV or reposition TX and/or RCV</td>
<td></td>
</tr>
<tr>
<td>Intermittent AF and RF signals</td>
<td>average</td>
<td>obstructions in signal path and/or RCV</td>
<td>remove obstructions or reposition TX</td>
<td></td>
</tr>
<tr>
<td>Intermittent AF and RF signals</td>
<td>average</td>
<td>squelch set too high</td>
<td>decrease squelch setting</td>
<td></td>
</tr>
<tr>
<td>Intermittent AF and RF signals</td>
<td>average</td>
<td>very strong RFI</td>
<td>identify source and eliminate, or change frequency of wireless microphone system</td>
<td></td>
</tr>
</tbody>
</table>

When multiple systems are in use, some additional problems can occur due to interaction between the systems. Turning individual systems on and off and trying systems in different combinations can help to pinpoint the cause. However, this can become much more difficult as the number of systems increases.

Following are some multiple system troubleshooting suggestions for symptoms observed when all systems are active.

<table>
<thead>
<tr>
<th>Conditions: TX on, RCV on, multiple systems</th>
<th>Symptom</th>
<th>Distance</th>
<th>Possible cause</th>
<th>Action</th>
</tr>
</thead>
<tbody>
<tr>
<td>Distortion on two (or more) systems with no AF peak indication</td>
<td>any</td>
<td>units on same frequency</td>
<td>change frequencies</td>
<td></td>
</tr>
<tr>
<td>Distortion on one (or more) systems with no AF peak indication</td>
<td>TX-TX short</td>
<td>TX + TX intermod</td>
<td>change frequencies</td>
<td></td>
</tr>
<tr>
<td>Distortion on one (or more) systems with no AF peak indication</td>
<td>TX-RCV short</td>
<td>TX + TX + RCV intermod</td>
<td>increase TX to TX distance</td>
<td></td>
</tr>
<tr>
<td>Distortion on one (or more) systems with no AF peak indication</td>
<td>TX-TX short</td>
<td>TX + TX intermod</td>
<td>change frequencies</td>
<td></td>
</tr>
<tr>
<td>Distortion on one (or more) systems with no AF peak indication</td>
<td>TX-RCV short</td>
<td>TX + TX + RCV intermod</td>
<td>increase TX to RCV distance</td>
<td></td>
</tr>
</tbody>
</table>
CHAPTER FIVE
APPLICATION NOTES

Following are some suggestions on wireless microphone system selection and use for some specific applications. Each section gives typical choices and setup for microphones, transmitters and receivers as well as a few operating tips.

Presenters

The most common wireless choice for presenters has been the lavaliere/bodypack system, which allows hands-free use for a single speaking voice. However, the traditional lavaliere microphone is often being replaced by a small headworn microphone because of its much better acoustic performance.

In either case, the microphone is connected to a bodypack transmitter and the combination is worn by the presenter. The receiver is located in a fixed position.

The bodypack transmitter is generally worn at the waistband or belt. It should be located so that the antenna can be freely extended and so that the controls can be reached easily. Transmitter gain should be adjusted to provide suitable level for the particular presenter.

The receiver should be located so that its antennas are line of sight to the transmitter and at a suitable distance, preferably at least 5 meters. Once the receiver is connected to the sound system the output level and squelch should be adjusted according to the previous recommendations.

The most important factor in achieving good sound quality and adequate gain-before-feedback with a lavaliere system is microphone choice and placement. A high quality microphone placed as close as practical to the wearers’ mouth is the best starting point. An omnidirectional lavaliere microphone should be attached to the presenter on a tie, lapel or other location within 8-10 inches of the mouth for best pickup.

The headworn microphone has a significant advantage because of its much closer placement to the mouth. Compared to a lavaliere microphone at 8 inches from the mouth, a headworn type placed within one inch of the mouth will have 18dB better gain-before-feedback. In addition, because the microphone is always at the same distance from the mouth, there is no volume change or tonal change as the presenter’s head moves in any direction.

In situations of limited gain-before-feedback or high ambient noise levels a unidirectional microphone may be used.

This type should be located like the omnidirectional type but it must also be aimed at the presenter’s mouth. The user should be aware that unidirectional types are much more sensitive to wind noise and breath blasts (k’s, t’s, d’s, etc.) as well as noise from clothing rubbing against the microphone or cable. Unidirectional microphones should always be used with a windscreen and mounted in a way to reduce direct mechanical contact with clothing or jewelry. Again, the headworn type has an advantage because the microphone itself is not in contact with clothing or other articles.

Finally, it should be noted that the unidirectional gain-before-feedback improvement is typically only 6-8dB. Thus, an omnidirectional headworn microphone will still have at least a 11-12dB advantage in gain-before-feedback over a unidirectional lavaliere type. This is sufficient to allow the use of omnidirectional headworn microphones in all but the most severe feedback environments. A unidirectional headworn microphone can perform nearly identically to a unidirectional handheld type and substantially better than any lavaliere type in this case.

Musical Instruments

The most appropriate choice for an instrument wireless application is a bodypack system, which will accept the audio signal from various instrument sources. The receiver can be a diversity design for highest performance or non-diversity for economy applications and is located in a fixed position.

The transmitter can often be attached to the instrument itself or to the instrument strap. In any case it should be located to avoid interfering with the player but with its controls accessible. Instrument sources include electric guitars and basses as well as acoustic instruments such as saxophones and trumpets. Electric sources can usually connect directly to a transmitter while acoustic sources require a microphone or other transducer.

Receivers for instrument systems are connected to an instrument amplifier for electric guitars and basses or to a mixer input for acoustic instruments, which are not otherwise amplified. Be aware of the potential for interference from digital effects processors in the vicinity of the amplifier or at the mixer position. Connections should be well-shielded and secure. Again the usual distance and line-of-sight considerations apply.
The most important factor in the performance of an instrument system is the interface between the instrument and the transmitter. The signals from electric instruments fitted with magnetic pickups are generally comparable to microphone signals, though the levels and impedances may be somewhat higher. Other transducers such as piezo-electric types have output signals that also are similar to microphone signals but again may have higher levels and substantially higher impedances. With any of these sources care should be taken to insure that there is compatibility with the transmitter input in regard to level, impedance and connector type.

Occasionally it is found that certain wireless microphone systems do not initially work well with certain instruments. Symptoms may include poor frequency response, distortion or noise. In most cases this can be traced to an impedance or level mismatch between the two. Frequency response changes are most often due to impedance problems. Make sure that the transmitter has sufficiently high input impedance. Distortion is usually due to excessive input level to the transmitter. Instruments with active circuitry (battery powered preamps) often have very high output levels which may need to be attenuated for some transmitters. They may also suffer fromRFI caused by the wireless microphone system. This may reduced by the addition of RF filters in the instrument.

A common type of noise that is heard in wireless microphone systems is often called modulation noise. This is a low-level hiss, which accompanies the actual instrument sound. Though it is usually masked by the instrument sound certain factors may make it more pronounced. These include low audio signal levels, low RF signal levels and high RF noise levels. Modulation noise can be most noticeable when the wireless microphone system is connected to a high gain instrument amplifier with boosted high frequencies and distortion circuits engaged. The apparent level of modulation noise can be reduced by setting the transmitter gain as high as possible (without causing distortion), maintaining adequate RF signal level and avoiding sources of RF noise.

Some electric guitars and basses used with wireless microphone systems may also exhibit intermittent noise when their control pots are moved to or from the endpoints of their rotation (full-on or full-off). This is due to metal-to-metal contact, which occurs at these points in certain potentiometer designs. A different type of pot may need to be substituted.

Microphones for acoustic instruments may be omni- or unidirectional and are usually condenser types. Microphone selection and placement for acoustic instruments is a subjective process that may involve a certain amount of trial and error. See the references in the bibliography for suggestions.

It is advised to consult the manufacturer of the wireless equipment and/or the manufacturer(s) of the instruments, microphones and transducers if problems persist. They may have details of suggested modifications for one or both units.

One wireless benefit of interest to guitar players is the elimination of the potential shock hazard created between a wired electric guitar and a wired microphone. Once the hardwire connection between either the guitar and amplifier or between the microphone and the PA system is removed the polarity of the guitar amp is of no consequence.

**Vocalists**

The usual choice for vocalists is a handheld wireless microphone system for close pickup of the singing voice. It consists of a suitable vocal microphone element attached to a handheld transmitter used with a fixed receiver.

The microphone/transmitter may be handheld or mounted on a microphone stand. Microphone technique is essentially the same as for a wired microphone: close placement gives the most gain-before-feedback, the least ambient noise pickup and the most proximity effect. An accessory pop filter may be used if wind or breath blast is a problem. If the transmitter is equipped with an external antenna avoid placing the hand around it. If the transmitter has externally accessible controls it may be useful to conceal them with a sleeve or tape to avoid accidental switching during a performance. Some transmitters can be set to lock out the controls. Battery condition should be checked prior to this if the indicator will be covered. Transmitter gain should be adjusted for the particular vocalist at performance levels.

A popular option for vocalists who require hands-free operation is the unidirectional headworn microphone. It can have gain-before-feedback performance equivalent to a handheld and similar sound quality as well. The only operational difference is that the vocalist cannot “work” the microphone by changing its distance from the mouth. Thus, vocal dynamics need to be adjusted with the singer's vocal technique rather than by microphone technique.

The receiver should be located at a suitable distance and in line of sight to the transmitter. Since this is often at the mixer position, check for possible interference from nearby digital signal processors. Again antenna and audio connections should be well-shielded and secure.
Selection and Operation of WIRELESS Microphone Systems

The primary considerations for sound quality in a hand-held wireless microphone system is the microphone element and its proper integration with the transmitter. The choice of element for a wireless microphone system would be made according to the same criteria as for a wired microphone. Ideally the wireless version of a microphone will sound identical to the wired version. Ultimately this is up to the manufacturer of the wireless microphone system. For this reason it is highly recommended to compare the performance of the proposed wireless microphone system to its wired counterpart to make sure that any differences in sound quality or directionality are minimal.

Aerobic/Dance Instruction

Aerobic and dance applications most often require bodypack wireless microphone systems to allow hands-free use by the instructor. The microphone is most often a headworn type, with a unidirectional element. This will give the best results for feedback control and overall sound quality. A lavaliere microphone may be used if gain-before-feedback is not a problem but it will not have the same sound quality as a headworn type. The receiver may be diversity or non-diversity depending on the performance level required and is located in a fixed position.

The transmitter is worn at the waist and must be securely attached since the user is generally quite active. The antenna should be freely extended and the controls accessibly located. Gain should be adjusted for the individual under actual use conditions.

The receiver should be located with the usual regard to distance and line of sight. In addition it should be out of the way of moving bodies and feet. Since these systems are often set up and taken down the connections should be regularly checked.

The primary criterion for aerobic/dance systems is reliability under extremely active conditions. These conditions include vibration, impact, heat, humidity and various bodily fluids! The basic system must certainly be durable but there are some additional steps that can be taken to improve long-term reliability. An accessory belt or pouch made of neoprene can protect the transmitter from mechanical damage as well as perspiration. It also provides a pad between the wearer and the transmitter to improve comfort and allows the transmitter to be easily repositioned if the instructor is doing floor exercises, etc. A design that covers the controls and/or connectors will further resist corrosion damage at these points of entry.

Microphone cable life can be prolonged by routing the cable to avoid extreme flexing or pull. Allow slack at the headband and at the point of entry to the transmitter. A side entry design provides more strain relief and allows for a drip loop in the cable to prevent perspiration from flowing down the cable into the connector. If an adjustable headband is used it should be adjusted only as often as necessary to maintain adequate stability.

The microphone element can be somewhat protected by using a foam windscreen. Periodically remove the windscreen, sponge it in warm soapy water, rinse and let dry. Replace when it shows signs of wear. However, even with these precautions the microphone and cable assembly in such a system should probably be considered a consumable item.

Theater

Theatrical applications also generally call for lavaliere/bodypack wireless microphone systems. The microphone and transmitter are worn by the performer while the receiver is in a fixed location. Theater combines aspects of presenter, vocalist, and aerobic/dance applications with additional unique requirements. In current theater practice the lavaliere microphone is often concealed somewhere on the head of the performer; just in front of the ear, on the forehead, in the hair or beard, etc. In some cases it is concealed in some part of the costume such as a hat or high collar. The intent is always to get the microphone as close to the performer’s mouth as possible without being visible. The close placement maximizes gain-before feedback and minimizes noise and acoustic interference. Miniature omnidirectional types are used almost exclusively, but they must be of high quality for both speech and singing. Avoid obstructing the ports on microphones with makeup or adhesives.

Headworn microphones have become much more common in theatrical applications, particularly for high-sound-level musical theater. Again, the benefits of very high gain-before-feedback, high signal-to-noise ratio, and consistent microphone-to-mouth distance make the headworn type an excellent choice in this setting.

Transmitters are also concealed in or under costumes and are often subject to an even more severe environment than the aerobic/dance situation. Special packs and bindings are available to attach the transmitter to various parts of the body. Latex covers are sometimes used to protect transmitters from sweat. Routing microphone cables and antennas and still allowing quick costume changes presents a serious challenge. Normal wear and tear on cables and connectors will take a rapid toll on anything but the most reliable microphones and transmitters.
Receivers for theatrical applications are not unique but they must be of high quality to allow multiple system use without interference. It is not unusual to use as many as 30 simultaneous wireless microphone systems in a professional musical theater production. This number can only be handled with systems operating in the UHF range. 10 to 12 systems is the practical limit at VHF frequencies. In addition, separate antennas and antenna distribution systems are necessary for any installation involving a large number of systems.

Though small-scale theater applications can be done with a moderate investment in planning and equipment, large-scale productions usually require professional coordination of wireless microphone systems to achieve successful results. This becomes an absolute necessity for a touring production.

**Worship**

Worship services may include presenter, vocalist and instrument applications. While wireless vocal and instrument use is essentially the same as outlined in the preceding sections, the presenter function may be somewhat different. Microphone, transmitter and receiver selection are as before but placement of the components may require extra consideration.

In particular, proper location of the lavaliere microphone and/or transmitter may pose problems because of robes or vestments. It is still necessary to position the microphone as close as practical the user’s mouth for best results. Different methods of attachment may be necessary. Access to transmitter controls can also be problematic. Use of accessory microphone mute switches similar to those worn by sports referees can be the answer. Though an omnidirectional type microphone is easier to use, a unidirectional model may be chosen to allow more gain-before-feedback. In this case pop sensitivity and mechanical noise should be taken into account. Again it is very important to adjust the transmitter level for the individuals’ voice under actual conditions.

Note that headworn microphones are becoming more acceptable for worship applications. They provide the highest gain-before-feedback in a hands-free type.

Because most worship services involve both wired lectern microphones and wireless lavaliere microphones it often happens that the person wearing the wireless is also speaking at the lectern. If the voice is picked up by both microphones an acoustic phenomenon known as “comb filtering” occurs which creates a hollow, unnatural sound. The solution is to turn down one of the two microphones whenever they are within one or two feet of each other. In most cases it will be less noticeable to turn down the lectern microphone when the wireless wearer approaches it.

Proper frequency selection is necessary for any worship application. Since a fixed location is the norm, unused TV channel frequencies are the best recommendation, not “traveling” frequencies. The simultaneous use of other wireless microphone systems by vocalists and musicians during the service must be considered as well. In addition, wireless microphone systems at other churches or facilities within 1000 feet of the site should be included in any program for frequency coordination.

Finally, receivers should be located and adjusted according to the suggestions made earlier. Even with proper squelch settings, though, it is very strongly recommended to turn off or turn down the outputs of any receivers that do not have an active transmitter. This will avoid noise from random RF interference being heard in the sound system.

**Conference/Corporate**

The growth of videoconferencing in meeting rooms and the increasing popularity of rooms with flexible seating layouts have made wireless microphones a popular option in corporate settings.

Wireless boundary or goose-neck microphones can provide excellent coverage of participants seated around a conference table or at smaller tables arranged classroom-style. The wireless form factor eliminates the need to drill holes in the table and allows the microphones to be put away when they are not needed.

Transmitters should be equipped with a convenient mute switch and an illuminated indicator that confirms the microphone status. Battery life should allow a full day of use, and rechargeable transmitters than can be stored in a charging station are convenient and intuitive for non-technical users. Robust standards-based encryption (such as AES-128 or AES-256) is often desired by corporate wireless users who desire privacy when confidential business is discussed.

Receivers mounted in an equipment rack should be connected to remote antennas that are in the meeting room itself. New antenna designs that resemble a wireless access point blend into the room décor more easily. A digital audio network connection (such as Dante) may make connecting to a digital audio processor simpler.

Spectral efficiency is important if many conference rooms in a facility will be equipped with wireless microphones. Some systems are capable of operating in a “high density” mode that enables many systems to fit in just one or two vacant TV channels. The 1.9 GHz DECT band is also a good option in the U.S. and many other countries, and is outside the TV band which may be an advantage in urban areas where TV signals are strong.
Film/Videography

Film and videography applications usually call for lavaliere/bodypack wireless microphone systems to minimize the visibility of the microphone. Handheld transmitters may also be used when visual appearance is not an issue. However, the receivers may be either fixed or portable. A common choice is a camera-mounted receiver used with a camcorder. Microphone/transmitter selection and placement are as outlined in other sections. Directional microphones are useful to control ambient noise. Placement can be consistent with visual requirements but should be positioned as close as practical to the sound source. The overall audio quality is largely determined by microphone selection and placement.

An important area in the use of wireless microphone systems with video and film equipment is the electrical interface between them. The interface is specified in terms of level, impedance, configuration (balanced/unbalanced) and connector. While the output characteristics of wireless receivers are well specified and fairly standard, the audio input characteristics of video equipment are often unspecified and unique. This is especially true for consumer video camcorders. Professional video units are normally designed with standard professional audio inputs.

Most camcorders that have a built-in microphone also have an input jack for an external microphone. This is usually a 1/8" mini phone jack. If the wireless receiver is equipped with a microphone-level 1/4" phone jack output a simple adapter cable will generally suffice. If the receiver only has an XLR-type output some additional concerns arise. An adapter cable or transformer can be used as described in the receiver setup section above.

Stereo camcorders often use an 1/8" stereo (tip-ring-sleeve) external microphone jack. To deliver a monophonic wireless signal to both the left and right audio channels a mono-to-stereo adapter must be inserted at the input jack. Some camera-mount receivers include a special output cable designed to work in either mono or stereo inputs.

Certain camcorder microphone inputs also supply a DC bias voltage for condenser microphones. In this case a transformer or blocking capacitor may be required to isolate the output of the wireless receiver from the DC. Consult the camcorder literature or manufacturer for details.

Camcorders that do not have manual audio level controls are always equipped with an automatic gain control (AGC). Its function is to maintain a constant audio recording level by increasing the gain for weak signals and decreasing the gain for strong signals. This circuit is generally not defeatable. When using the built-in microphone most direct sound sources are at a significant distance. At the microphone the overall variation between direct sound level and background sound level is not very large.

Close-talked microphones, either handheld or lavaliere, present a much stronger signal relative to the background sound level. With these devices the AGC will operate at high gain when only background sound is present and will quickly reduce gain when a strong close-talk signal occurs. This will result in audible "pumping" of the background noise level. The same effect is typically heard in live sports broadcasts; the crowd noise is quickly suppressed when the announcer speaks and slowly returns when the announcer is silent. Unfortunately, if the AGC is not defeatable there is no convenient way to eliminate this effect. Operating the wireless microphone system at very low levels can force the AGC to operate at full gain but this will result in noisy audio tracks.

Frequency selection for film/videography should be done according to the guidelines already presented. One additional source of potential interference is the video equipment itself since it contains digital and RF circuitry. Listen for audible effects from both the transmitter and from the receiver when they are used close to the camera and/or video recorders.

Broadcast

Broadcast applications may require handheld, plug-on and/or bodypack wireless microphone systems. To interview random subjects most field reporters and roving talk show hosts prefer a handheld or plug-on transmitter for maximum mobility and control. Bodypack systems are used to pick up a single talent when a handheld type would be cumbersome. Receivers may be in a fixed location for studio use or may be portable for field use.

Omnidirectional microphones are the preferred choice for situations where ambient noise is not excessive. This allows more flexibility in placement and orientation as well as reduced sensitivity to wind and handling noise. When ambient noise is a factor or in a setup where feedback is a possibility a unidirectional microphone may be necessary. Microphones with good sound quality and durability are a must.
Lavaliere microphones will require a bodypack transmitter. If a desired handheld microphone model is available in a wireless version it can be used directly. Since most handheld transmitters use unidirectional microphone elements a plug-on transmitter may be necessary for use with handheld omnidirectional microphones or other wired hand-held models.

Even for portable or camera-mount use a diversity receiver is standard choice for professional broadcast applications. Antenna location should be well planned, especially when used in a studio environment with lighting grids and other large metal structures. Receivers should be located away from lighting controllers, digital audio equipment and video equipment that may produce interference. Balanced audio lines are also standard procedure.

Receivers used in remote trucks face interference not only from nearby audio and video equipment but they may also be subject to interference from two-way radio systems and remote-to-studio transmitters operating on VHF or UHF frequencies. Two-way transceivers should not be operated close to wireless transmitters or receivers. The wireless microphone system receivers should also be located well away from remote link transmitters. In addition, both the wireless microphone system antennas and the remote link antennas should be properly installed for minimum interference with each other.

Frequency selection for broadcast involves the same considerations as in other applications for studio use. In ENG (Electronic News Gathering) or EFP (Electronic Field Production) situations the additional factors of nearby remote transmitters and the possibility of wireless microphone systems in use by other broadcast crews must be taken into account. In a local area it is sometimes possible to coordinate frequencies between different stations ahead of time to reduce the likelihood of frequency conflict at a news site. Specifying high quality frequency-agile wireless equipment will further minimize interference problems.

## General requirements for point-to-point wireless include:

- High fidelity audio
- Line level in and line level out
- AC-power capability for both transmitter and receiver
- "Sufficient" transmission distance

Most radio communication products such as wireless intercoms, two-way radios, and mobile phones are unsuitable due to lack of audio fidelity. However, two common high-fidelity radio products that can be used in a point-to-point application are the wireless microphone system and the wireless in-ear monitor system. Each of these has slightly different concerns/advantages in performance and in setup.

### Using wireless microphones for point-to-point

A bodypack wireless system can fulfill the requirements above with two modifications. (See Figure 5-1.) First, though many receivers have line level outputs, few transmitters are capable of accepting a line level input signal. This can be accomplished by using an external pad or attenuator at the transmitter input. Second, though most receivers run on AC power, transmitters are battery-only devices. Using a suitable external DC power supply can provide the necessary AC capability for the transmitter.

Once the input signal is sufficiently attenuated and power is provided, the bodypack transmitter should be located to provide the best line-of-sight to the receiver. This usually involves securing the pack to a pole or some other elevated structure. If wet weather is a factor, the transmitter can be protected by a “ziplock” bag or other covering.

At the receiver, no modification is usually required since it is AC powered and has a line level output.
Using wireless in-ear monitors for point-to-point

A wireless in-ear monitor system (IEM) can also be used as a point-to-point system with only one modification at the receiver. At the transmitter, no modification is usually required since it is AC powered and can accept line level signals directly. (See Figure 5-2.) The modification to the receiver is again for AC power. Many IEM receivers use 9-volt batteries and thus can use a procedure similar to that for the 9-volt transmitters above. The only other accommodation necessary at the receiver is adapting the stereo (TRS) mini-phone jack output to connect to the destination audio system. This is effectively an unbalanced, -10dBv signal, suitable for most line level input devices. It is recommended that any receiver limiter be switched off for point-to-point applications in order to obtain the maximum drive level.

Once power and audio connections are made, the IEM receiver should be secured to an elevated location to allow best line of sight to the IEM transmitter. Again, weather protection for the receiver can be provided by a ziplock bag or similar covering.

A significant difference between IEM and wireless microphones for this application is that the stereo IEM can transmit two audio channels (multiplexed) per radio frequency while the wireless microphone system can transmit only one audio channel per radio frequency. If stereo transmission is required IEM is more cost- and spectrum-efficient.
Combining in-ear monitors with wireless microphones for point-to-point

It is sometimes possible to use an IEM transmitter in conjunction with a wireless microphone receiver to create a point-to-point system that is already set up at each end for line-level audio signals and AC power. The principle requirements are that the operating frequencies match and that the audio companding schemes are compatible. Of course only mono audio signals may be transmitted due to the mono nature of the wireless microphone receiver.

The one modification that may be needed is to defeat the squelch circuit of the wireless microphone receiver. Certain IEM transmitters can now be used with compatible wireless microphone receivers without defeating tone-key squelch in the receiver. This is accomplished by a menu setting in the IEM transmitter which matches its tone-key frequency to that of the receiver.

Finally, it may be possible to use a wireless microphone transmitter with an IEM receiver to create an entirely battery-powered point-to-point system. Again, the radio frequencies and companding schemes must match. However, it is probably more efficient to use a wireless microphone with a matching camera-mount type receiver if a completely battery-powered system is the goal.

Antennas

The antennas supplied with most wireless products are omnidirectional. These are suitable for both point-to-point and point-to-multi-point applications. If additional transmission range is required it may be possible to use directional receiving and/or transmitting antenna(s). Note that although diversity receivers are always preferred it may not be necessary (or practical) to use directional antennas on both antenna inputs for point-to-point. Since the transmitter and receiver locations are assumed to be fixed in this application, multipath variations should be minimal once the equipment is set up.

Maximum range of point-to-point wireless

The practical range for most of these systems in point-to-point applications is comparable to their published range in normal usage. For standard wireless systems this ranges from about 150 ft. to 800 ft. depending on various conditions. Directional antennas may boost these ranges by 50%. Good line of sight and the normal precautions for frequency selection are assumed.

The maximum range system employs an IEM transmitter with either an IEM receiver or a suitable wireless microphone receiver. The range advantage of these systems is due to two factors: first, the transmitter power is higher, up to 100mW. This is at least twice the power of most wireless microphone systems. Second, since both the transmitter and the receiver may have detachable antennas, it is possible to use a directional antenna on both the transmitter and the receiver.

In operation, the receiving and transmitting antennas should be pointed toward each other and oriented vertically. Elevation for best line of sight will further improve range. Such a system may be capable of stereo transmission up to 2500 ft and mono operation up to 3500 ft.

If greater distances are required, it can be accomplished by using an additional system as “repeater.” That is, at the location of the first receiver a second transmitter is set up to rebroadcast the signal to a more distant second receiver. Of course, each of these additional systems has to be on a different compatible frequency. The practical limit for a repeater system using standard wireless equipment is about 3 “hops” due to increased noise and distortion.

Summary

It is possible to employ wireless systems for point-to-point applications in several ways. Both wireless microphone systems and IEM systems can be used, each with different adaptations. The modifications are primarily for AC powering of battery powered devices and for matching audio signal levels. Special connectors and/or adapters may also be necessary in some cases.

For moderate distance, single channel applications may use wireless microphone systems for good results. Multi-channel, especially stereo, transmission may benefit from the IEM approach. Longer distance uses will require directional antennas for wireless microphone receivers or for IEM transmitters. Maximum range applications can be handled by the dual directional antenna IEM setup or by using multiple systems in a repeater configuration.
Conclusion

To the extent that wireless audio technology is a work in progress, this presentation is also a work in progress. As new radio technologies develop, new applications will arise to take advantage of them. At the same time, new spectrum rules and allocations will likely challenge the capability of wireless equipment suppliers and users to provide successful solutions for those applications.

Implementing successful solutions will rely not just on the programmed "intelligence" built into the equipment but also on the dynamic intelligence of the people specifying and using the equipment. That intelligence comes both from study of the principles underlying the technology and from experience gained by using the technology in real-world situations.

It is hoped that the material presented in this publication will serve as a useful reference on the technical principles of wireless audio transmission, as well as a practical guide to using those principles in a wide variety of applications.
**Calculation of Intermodulation Products**

The simplest IM products that can occur between any two operating frequencies \(f_1\) and \(f_2\) are the sum of the two frequencies and the difference between the two frequencies:

\[
\begin{align*}
  f_1 + f_2 & \quad \text{(sum)} \\
  f_1 - f_2 & \quad \text{(difference)}
\end{align*}
\]

If we choose \(f_1 = 200\) MHz and \(f_2 = 195\) MHz, then:

\[
\begin{align*}
  f_1 + f_2 &= 200 + 195 = 395\text{ MHz (sum)} \\
  f_1 - f_2 &= 200 - 195 = 5\text{ MHz (difference)}
\end{align*}
\]

These IM products are sufficiently far away from the original frequencies that they will generally not cause problems to a third wireless microphone system in the original frequency band.

However, as mentioned earlier, other products of non-linear circuits are multiples of the original frequency. That is, application of a single frequency to a non-linear circuit will produce additional products that are double, triple, quadruple, etc. the original frequency. Fortunately, the strength of these products decreases rapidly as the order (multiplication factor) increases. The practical result is that only the products at two times and three times the original frequency are significant. Since these products then combine as sums and differences with themselves and with the original frequencies, the following additional products can occur:

\[
\begin{align*}
  (2 \times f_1) \\
  (2 \times f_2) \\
  (3 \times f_1) \\
  (3 \times f_2) \\
  (2 \times f_1) \pm f_2 \\
  (2 \times f_2) \pm f_1 \\
  (3 \times f_1) \pm f_2 \\
  (3 \times f_2) \pm f_1 \\
  (2 \times f_1) \pm (2 \times f_2) \\
  (3 \times f_1) \pm (2 \times f_2) \\
  (3 \times f_2) \pm (2 \times f_1) \\
  (3 \times f_1) \pm (3 \times f_2)
\end{align*}
\]

The "order" or type of IM product is identified by the particular combination of frequencies that created it. The order of an IM product is the sum of the multipliers (coefficients) of the frequencies in the expressions above. The complete group of possible frequencies (original frequencies, intermodulation products and combinations) that can exist when two systems (at 200 MHz and 195 MHz for this example) are operated simultaneously is thus:

<table>
<thead>
<tr>
<th>Two-Transmitter Intermodulation Calculation</th>
<th>Order</th>
<th>Frequency</th>
<th>Significant</th>
</tr>
</thead>
<tbody>
<tr>
<td>(f_1) (original frequency)</td>
<td>1</td>
<td>200</td>
<td>Yes</td>
</tr>
<tr>
<td>(f_2) (original frequency)</td>
<td>1</td>
<td>195</td>
<td>Yes</td>
</tr>
<tr>
<td>(2 \times f_1)</td>
<td>2</td>
<td>400</td>
<td>No</td>
</tr>
<tr>
<td>(2 \times f_2)</td>
<td>2</td>
<td>390</td>
<td>No</td>
</tr>
<tr>
<td>(f_1 + f_2)</td>
<td>2</td>
<td>395</td>
<td>No</td>
</tr>
<tr>
<td>(f_1 - f_2)</td>
<td>2</td>
<td>5</td>
<td>No</td>
</tr>
<tr>
<td>(3 \times f_1)</td>
<td>3</td>
<td>600</td>
<td>No</td>
</tr>
<tr>
<td>(3 \times f_2)</td>
<td>3</td>
<td>585</td>
<td>No</td>
</tr>
<tr>
<td>((2 \times f_1) + f_2)</td>
<td>3</td>
<td>595</td>
<td>No</td>
</tr>
<tr>
<td>((2 \times f_1) - f_2)</td>
<td>3</td>
<td>205</td>
<td>Yes</td>
</tr>
<tr>
<td>((2 \times f_2) + f_1)</td>
<td>3</td>
<td>580</td>
<td>No</td>
</tr>
<tr>
<td>((2 \times f_2) - f_1)</td>
<td>3</td>
<td>190</td>
<td>Yes</td>
</tr>
<tr>
<td>((3 \times f_1) + f_2)</td>
<td>4</td>
<td>795</td>
<td>No</td>
</tr>
<tr>
<td>((3 \times f_1) - f_2)</td>
<td>4</td>
<td>405</td>
<td>No</td>
</tr>
<tr>
<td>((3 \times f_2) + f_1)</td>
<td>4</td>
<td>785</td>
<td>No</td>
</tr>
<tr>
<td>((3 \times f_2) - f_1)</td>
<td>4</td>
<td>385</td>
<td>No</td>
</tr>
<tr>
<td>((2 \times f_1) + (2 \times f_2))</td>
<td>4</td>
<td>790</td>
<td>No</td>
</tr>
<tr>
<td>((2 \times f_1) - (2 \times f_2))</td>
<td>4</td>
<td>10</td>
<td>No</td>
</tr>
<tr>
<td>((3 \times f_1) + (2 \times f_2))</td>
<td>5</td>
<td>990</td>
<td>No</td>
</tr>
<tr>
<td>((3 \times f_1) - (2 \times f_2))</td>
<td>5</td>
<td>210</td>
<td>Yes</td>
</tr>
<tr>
<td>((3 \times f_2) + (2 \times f_1))</td>
<td>5</td>
<td>985</td>
<td>No</td>
</tr>
<tr>
<td>((3 \times f_2) - (2 \times f_1))</td>
<td>5</td>
<td>185</td>
<td>Yes</td>
</tr>
<tr>
<td>((3 \times f_1) + (3 \times f_2))</td>
<td>6</td>
<td>1185</td>
<td>No</td>
</tr>
<tr>
<td>((3 \times f_1) - (3 \times f_2))</td>
<td>6</td>
<td>15</td>
<td>No</td>
</tr>
</tbody>
</table>

Though this list of calculated frequency combinations is lengthy, it can be seen that only the IM products at 185, 190, 205 and 210 MHz are in the same general band as the two original operating frequencies. These products will not cause compatibility problems between the two original systems but can interfere with other systems that may be added in this band. In this example, the operating frequency of a third system should be chosen to avoid these four IM frequencies. In general, only odd-order IM products are considered because even-order products typically fall well away from the original frequencies, as shown above. Furthermore, though higher odd-order IM products may also fall near the original frequencies, only 3rd order and 5th order IM products are strong enough to be of concern.

If three or more systems are operated simultaneously, the situation becomes somewhat more complicated but the same principles apply. In addition to the IM products calculated for each pair of frequencies, products due to combinations of three transmitters must also be considered.
In this example, the third system frequency (187 MHz) has been chosen to avoid the first two frequencies and their respective IM products. A third system that coincided with an IM product may experience interference when its transmitter is far from its receiver while the first two transmitters are close to each other and to the third receiver.

Note that the addition of the third frequency creates four new third-order, two-transmitter products as well as three third-order, three-transmitter products. In general, N transmitters create N x (N-1) third-order, two-transmitter IM products, as well as a number of third-order, three-transmitter products. Thus, the number of available frequencies for additional systems decreases exponentially as the number of systems increases. For this reason, computer programs are used to generate and evaluate compatible sets of frequencies.

5th order two-transmitter IM products are not usually strong enough to cause problems, but may be a factor in cases of extreme transmitter or receiver proximity. 5th order three-transmitter IM products such as \((3 \times f_1) - (2 \times f_2) - f_3\) and \((2 \times f_1) - (2 \times f_2) + f_3\) are generally too weak to be of concern.

Maintaining adequate physical distance between transmitters and between transmitters and receivers will minimize the creation of IM products. The figure below indicates the effect of distance on the amplitude of 3rd order IM products created by two transmitters.
US Television Channels

<table>
<thead>
<tr>
<th>Channel</th>
<th>Frequency Range (MHz)</th>
</tr>
</thead>
<tbody>
<tr>
<td>2</td>
<td>54 – 60</td>
</tr>
<tr>
<td>3</td>
<td>60 – 66</td>
</tr>
<tr>
<td>4</td>
<td>66 – 72</td>
</tr>
<tr>
<td>5</td>
<td>76 – 82</td>
</tr>
<tr>
<td>6</td>
<td>82 – 88</td>
</tr>
<tr>
<td>7</td>
<td>174 – 180</td>
</tr>
<tr>
<td>8</td>
<td>180 – 186</td>
</tr>
<tr>
<td>9</td>
<td>186 – 192</td>
</tr>
<tr>
<td>10</td>
<td>192 – 198</td>
</tr>
<tr>
<td>11</td>
<td>198 – 204</td>
</tr>
<tr>
<td>12</td>
<td>204 – 210</td>
</tr>
<tr>
<td>13</td>
<td>210 – 216</td>
</tr>
<tr>
<td>14</td>
<td>470 – 476</td>
</tr>
<tr>
<td>15</td>
<td>476 – 482</td>
</tr>
<tr>
<td>16</td>
<td>482 – 488</td>
</tr>
<tr>
<td>17</td>
<td>488 – 494</td>
</tr>
<tr>
<td>18</td>
<td>494 – 500</td>
</tr>
<tr>
<td>19</td>
<td>500 – 506</td>
</tr>
<tr>
<td>20</td>
<td>506 – 512</td>
</tr>
<tr>
<td>21</td>
<td>512 – 518</td>
</tr>
<tr>
<td>22</td>
<td>518 – 524</td>
</tr>
<tr>
<td>23</td>
<td>524 – 530</td>
</tr>
<tr>
<td>24</td>
<td>530 – 536</td>
</tr>
<tr>
<td>25</td>
<td>536 – 542</td>
</tr>
<tr>
<td>26</td>
<td>542 – 548</td>
</tr>
<tr>
<td>27</td>
<td>548 – 554</td>
</tr>
<tr>
<td>28</td>
<td>554 – 560</td>
</tr>
<tr>
<td>29</td>
<td>560 – 566</td>
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<td>602 – 608</td>
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<tr>
<td>37</td>
<td>608 – 614 (reserved for radio astronomy &amp; medical telemetry)</td>
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38 614 – 620
39 620 – 626
40 626 – 632
41 632 – 638
42 638 – 644
43 644 – 650
44 650 – 656
45 656 – 662
46 662 – 668
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48 674 – 680
49 680 – 686
50 686 – 692
51 692 – 698

Reallocated in 700 MHz Band Auction (2010)
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54 710 – 716
55 716 – 722
56 722 – 728
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61 752 – 758
62 758 – 764
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69 800 – 806

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<tr>
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<td>Washington, DC</td>
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Technical Considerations for Digital Wireless Audio Systems

In concept, fully digital wireless transmission is simple. Add an analog-to-digital (A/D) converter at the input of the transmitter. Transmit the resulting digital information to the receiver. Demodulate the digital information and add a complementary digital-to-analog (D/A) converter at the output of the receiver. The ultimate limitation lies in the amount of digital information that must be reliably transmitted for acceptable audio quality.

In general, information transmission techniques (wired or wireless) must balance bandwidth limitations with hardware (and software) complexity. Bandwidth refers to the range of frequencies and/or amplitudes used to convey the information. In audio, a frequency range of 20-20,000Hz and an amplitude range (dynamic range) of 120dB is perhaps the ultimate goal. However, a frequency range of 300-3000Hz and a dynamic range of 30dB are sufficient for telephone-quality speech. As expected, high fidelity audio equipment tends to be more complex and costly than telephone equipment.

In analog FM radio systems, audio fidelity is greatly dependent on allowable deviation, which is related to RF bandwidth: wider deviation increases occupied bandwidth. Regulatory bandwidth limitations necessitate the use of companders to achieve acceptable dynamic range in most high quality analog wireless systems.

The bandwidth required for a high fidelity digital wireless system depends on the amount of digital information transmitted and the transmission rate. In practice, the bandwidth of digital systems is also limited by physical and regulatory requirements. Ultimately, the fidelity and reliability of a digital wireless system is limited by these same bandwidth restrictions.

A digital representation of an analog signal is generated by sampling the analog waveform at some rate. To avoid errors in the digital representation of an audio signal, the sampling rate must be at least twice the highest audio frequency desired. Thus, digitizing a 20 kHz audio signal requires a sampling rate of at least 40kHz. CD audio uses a 44.1kHz sampling rate while advanced systems use sampling rates of 48kHz or higher.

The measurement of each sample is represented by a digital number. The resolution (accuracy) of this value is proportional to the number of "bits" in the digital number, where each bit can be "0" or "1". The greater the number of bits, the more accurate is the sampled value. For example, an 8-bit sample represents 256 values, a 16-bit (CD quality) sample has 65,536 values, and a 24-bit sample represents more than 16 million values.

The bit rate of a digital signal is the resolution multiplied by the sampling rate. CD audio is 16 bits x 44.1 kHz or 705.6k bits-per-second (bps). Higher performance digital systems that operate at a resolution of 24 bits and a sampling rate of 48 kHz generate a bit rate of 1152kbps.

In the simplest form of digital audio transmission, the required bandwidth of such a signal would be equal to the bit rate. That is, to transmit CD-quality audio would require a bandwidth of 705.6 kHz while a high performance signal would require 1152kHz. Based on allowable occupied bandwidth limits, it is not possible (or legal) to transmit such a signal. For comparison, cellular telephones use 8-bit resolution and a 6kHz sample rate, resulting in a data rate of only 48kbps. The resulting audio quality limitation is obvious.

To deliver the highest quality audio signal in a practical digital wireless system, it is necessary to fit the transmitted data within the allowable bandwidth limits using various data management techniques. These techniques are implemented by first "coding" the data in the transmitter and then "decoding" the transmitted data in the receiver. This process is conceptually similar to "companding" processes in analog FM systems, but here it is carried out in the digital domain. Properly implemented, the resulting audio can have wider frequency response, better dynamic range, and less audible "artifacts" than traditional analog systems. In addition, the digital transmission scheme can offer some improvement in radio performance as well as the possibility of data encryption.

The basic method of transmitting digital data in a radio signal involves modulating the carrier in discrete steps, a process called "shift keying". This can be done with frequency (Frequency Shift Keying or FSK), amplitude (Amplitude Shift Keying or ASK), or phase (Phase Shift Keying or PSK). If the shift key signal has two possible values (corresponding to a bit value of "0" or "1") then the data rate is equal to the bit rate. For example, a 2-PSK signal shifts between two possible phase values where the shift from the first value to the second might represent binary "0" and the shift from the second value to the first might represent binary "1". (See Figure C-1.)
However, it is possible to send more than one bit per cycle by coding multiple bits into "symbols". The symbol rate is equal to the bit rate divided by the number of bits transmitted with each symbol. The theoretical occupied bandwidth of such a coded digital transmission is now equal to the symbol rate. In this way it is possible to transmit the same amount of data in less bandwidth. For example, QPSK coding (Quadrature Phase Shift Keying) has four possible values (00, 01, 10, 11) and can thus send two bits per symbol, resulting in half the required bandwidth of a non-coded signal. (See Figure C-2.)

By combining modulation schemes, the symbol coding can have higher efficiency. For example, quadrature amplitude modulation (QAM) combines PSK with ASK to add a fourth bit to the symbol value. (See Figure C-4)

However, even with the reduced bandwidth requirements of higher order symbol coding techniques it is still not generally possible to reliably transmit the entire data stream within the available bandwidth. To reduce the data rate to manageable size some form of data compression must be employed.

Data compression schemes can be characterized as "lossless" or "lossy". A lossless scheme reduces the data stream by coding redundant data into shorter forms. Decoding "lossless" compressed data will exactly restore the entire original data stream. Lossless compression is effective on data that exhibits significant redundancy such as text, spreadsheets, and certain graphic files. It does not provide a high degree of compression on typical digital audio data.

A lossy data compression scheme discards a certain amount of the original data stream, generally the "least significant" parts. For audio data, there are various methods that rely on the characteristics and limitations of human audio perception. The best lossy compression schemes result in audio that is nearly indistinguishable from the original. However, each method involves tradeoffs between the degree of compression, audio quality, and processing time (latency).

The algorithm (mathematical formula) used in a particular data compression scheme is called a "codec" (code – decode), indicating the two-step nature of the process. Examples of codecs used in digital audio recording are MP3 and AAC. Each of these has selectable bit rates, the lowest offering small file size with reduced fidelity and the highest offering improved audio fidelity at the expense of larger file size.

However, these standard codecs are relatively slow, introducing latency (delay) of up to 100ms in the process.

Likewise, 8-PSK (Eight Phase Shift Keying) has eight possible values (000, 001, 010, 011, 100, 101, 110, 111) accommodating three bits per symbol, thus reducing the required bandwidth to one-third of a non-coded signal. (See Figure C-3)
In addition to the codec, other sources of latency include A/D and D/A converters, as well as other DSP elements. Though this may not be a problem in a music playback system, it would be unacceptable for a live performance wireless audio system. In live audio applications, latency must be kept as short as possible, ideally less than 5ms. Above 5ms, the latency becomes generally noticeable and when it exceeds 20ms, musical timing can be affected. For comparison, cellular telephone connections may have latency greater than 200ms. This leads to well-known effects such as unnatural pauses in conversation and the tendency for each person to speak over the words of the other.

Professional digital wireless systems use specialized codecs that are capable of excellent audio quality and low latency within allowable radio bandwidth limitations. For example, in the 900MHz band, the allowable bandwidth is about 500kHz. Assuming a simple 1-bit per symbol transmission, an 1152kbps data stream (48kHz x 24 bit) will require about 3:1 data compression (1152 / 3 = 384kbps). In the standard UHF broadcast band, the allowable bandwidth is less than 200kHz. This necessitates higher symbol coding and/or increased data compression.

A typical transmit coding scheme involves digitally dividing the full audio bandwidth into a number of sub-bands. Each sub-band is processed separately to deliver the desired amount of data compression. The sub-band data are assembled into a digital data stream which is used to modulate the transmitter output. Depending on allowable occupied bandwidth and other limitations, the modulation may be FSK, PSK, or higher level coding schemes such as QAM.

In the matching receiver, the incoming data stream is disassembled into corresponding sub-band data. Each sub-band is decoded to reproduce its original counterpart. Finally, the decoded sub-band data are combined to reconstruct the original full-bandwidth audio signal. It should be noted that the parameters for each element in this process have to be very carefully chosen to achieve the highest audio quality and the lowest latency consistent with both hardware and software limitations.

One necessary addition to the data stream is control and timing information to achieve synchronization between the transmitter and receiver processes. Other transmitted data may include battery condition, transmitter type, gain settings, and data encryption settings. A certain amount of bandwidth is always required for these functions, further limiting the maximum data rate.

The final consideration for digital systems is data transmission error. The integrity of the transmitted data stream can be adversely affected by radio interference, multi-path, signal fading, etc. The likelihood of errors also increases with the complexity of the transmission scheme. (See Figure C-5)

To reduce the effects of data loss, various error management techniques are employed. Some of these involve transmitting extra data to provide not only a means of detecting data loss but also a means of correcting some of those losses. However, since error correction methods may increase occupied bandwidth as well as processing complexity/latency, complete error elimination is not possible. Ultimately, if the data error rate exceeds some maximum, the system will mute to prevent audible artifacts. Properly designed digital wireless system can be made very robust in the presence of transmission errors and able to recover gracefully from any severe data loss.
Absorption
The weakening of radio wave strength by losses in various materials

ADC
Analog-to-digital converter, also abbreviated A/D

AF
Audio frequencies, typically 20-20,000 Hz.

AM
Amplitude modulation

Ambient
Local or background, ie. ambient noise

Amplitude
Magnitude or strength of a signal or wave

AM rejection
Ability of an FM receiver to reject signals from AM transmitters and/or AM noise from electrical devices or natural sources

Antenna
Electrical circuit element that transmits or receives radio waves

Antenna gain
Measure of antenna efficiency compared to a reference antenna, ie. dBi = gain relative to theoretical isotropic (spherical omnidirectional) antenna

Antenna splitter
A device for electrically matching a single antenna to multiple receivers

ASK
Amplitude shift keying

Attenuation
Measure of the loss of amplitude of a signal

Band
A defined portion of the frequency spectrum

Bandwidth
A measure of the frequency range of a signal or device

Base frequency
The actual frequency of a crystal oscillator, usually then multiplied to some higher operating frequency

Bias voltage
A fixed DC voltage which establishes the operating characteristic of a circuit element such as a transistor

Bit
Smallest digital data element, has a value of ‘0’ or ‘1’

Bodypack
Transmitter style which can be worn on the body

Capture
The effect of a strong FM signal suppressing weaker signals at the receiver

Carrier
The basic or unmodulated radio wave

Cascade
Active antenna output port of receiver or antenna splitter than can feed another receiver or antenna splitter

Codec
Software algorithm for data compression and decompression (code-decode)

Compander
A two-step noise reduction system consisting of a compressor in the transmitter and an expander in the receiver

Compressor
A circuit which reduces the dynamic range of a signal by a fixed ratio, typically 2:1 in a compander system

DAC
Digital-to-analog converter, also abbreviated D/A

dBm
Ratio of power levels relative to one milliwatt, ie. 0dBm = 1mW, 10dBm = 10mW

DECT
Digital Enhanced Cordless Telecommunications, standard for digital cordless telephones, ie. 1920-1930 MHz in the US, different bands in other countries

De-emphasis
A fixed equalization which typically rolls off high frequencies in the second step of a two-step noise reduction process

Demodulation
The recovery of the original modulating information from a radio signal

Detector
The circuitry that performs demodulation

Deviation
The maximum frequency variation of an FM signal

Diffraction
The bending or partial reflection of radio waves by metal objects

Dipole
An antenna which is made up of two active elements

Direct
Not reflected

Direct sequence spreading
Spread spectrum technique using continuous, synchronized deviation changes across a wide frequency range

Distortion
Any unwanted difference between the original and final version of a signal

Diversity
Scheme that relies on statistical decorrelation of the same signal received at multiple locations to minimize signal fading and multipath dropout, also called space diversity or antenna diversity

Dropout
The complete loss of received signal due to multipath interference

DSP
Digital Signal Processor, used in digital and hybrid radio circuits

Dynamic range
Maximum amplitude range of a signal or device, generally the difference between the strongest and weakest signals that occur or that the device can handle

Encryption
Encoding a transmitted signal so that it can be decoded only with a unique digital “key”

EM
Electromagnetic

ERP
Effective radiated power, the actual power radiated by a transmitter antenna

Expander
A circuit which expands the dynamic range of a signal by a finite ratio, typically 1:2 in a compander system

Fading
Loss of signal strength due to distance or obstacles in the signal path

FCC
Federal Communications Commission, US radio regulatory agency

Field
A distribution of energy in space, ie. electric, magnetic, sound

Field strength
The amplitude of a field at a given point, measured in volts per meter for electric energy
FM
Frequency modulation

Frequency
A measure of the rate of variation of a wave or signal

Frequency agile
Having the ability to change frequencies: tunable

Frequency diversity
Scheme that relies on statistical decorrelation of the same signal transmitted and received at multiple frequencies to minimize RF interference effects

Frequency hopping
Spread spectrum technique using continuous, synchronized frequency changes across a wide frequency range

Frequency response
Variation in amplitude of a signal over a range of frequencies

Front end
Initial filter stage of a receiver

FSK
Frequency shift keying

Gain
Amplification

Geolocation
Ability of a TVBD to determine its geographic location

Ground plane
Electrical approximation of a zero-potential reflective surface at the base of an antenna

GSM
Global Systems for Mobile Communications, mobile telephone standard; ie. 850 Mz and 1900 MHz in the US, different bands in other countries

Handheld
Transmitter type which can be held in the hand

Hash
A term for audible radio interference

Helical
Wideband directional antenna consisting of a single circularly-polarized element

Heterodyne
To combine signals of various frequency in a manner that produces additional signals at frequencies which are sums and differences of the original frequencies

Hybrid
Radio design combining digital audio processing with analog transmission

IEM
In ear monitor

IF
Intermediate frequency, a lower radio frequency (typically 10.7 MHz) found in the middle stages of a receiver

IFB
Interruptible fold back, a cueing system used in broadcast

IF filter
A narrow, band-pass filter tuned to the intermediate frequency (IF)

IM
Intermodulation, frequencies produced by combinations of other frequencies in non-linear devices

IMD
Intermodulation distortion (another name for IM)

IM rejection
Ability of a receiver to reject IM products

Image
An interfering frequency which differs from the desired frequency by twice the intermediate frequency (IF)

Image rejection
The ability of a receiver to reject interference from an image frequency (determined by the front end)

Impedance
A measure of the resistance to current flow in a circuit (may vary with frequency)

Indirect
Reflected or diffracted

Intercept (third order)
A measure of the ability of a radio input stage to handle high signal levels without overload or distortion

Inverse square law
Mathematical relationship in which one quantity is inversely proportional to the square of another quantity, ie. signal strength decreases according to the distance squared

Latency
Signal delay that occurs in digital systems due to finite speed of DSP, ADC, and DAC circuits

Level
The amplitude or strength of a signal

LF
Low frequency

Limiter
A circuit which limits the maximum level of a signal

Line-of-sight
Preferred wireless operating condition where transmit antenna is visible to receive antenna with no obstruction

LO
Local oscillator, in a receiver it is tuned to a frequency which is offset from the operating frequency by an amount equal to the intermediate frequency (IF)

Log periodic
Wideband directional antenna consisting of multiple logarithmic-spaced dipole elements

Loss
Decrease in signal strength during transmission, propagation or reception

Medium
Substance through which a wave propagates; for radio it may be vacuum, gas, liquid or solid, wave speed is affected by medium

Mixer
Circuitry in a receiver that combines the received signal with the local oscillator to produce the IF signal

Modulating frequency
The frequency of the audio signal used for modulation of the radio wave

Modulation
Variation of a wave parameter (such as amplitude or frequency) to carry information

Modulation noise
Low level noise which accompanies the audio signal in a companded wireless microphone system

Multipath
Reflection of radio waves that causes fluctuation in received signal strength

Narrow band
An FM signal in which the deviation is much less than the modulating frequency
Noise
Undesirable random audio or radio energy

Operating frequency
The final output frequency of a transmitter or the tuned frequency of a receiver

Oscillator
A circuit that produces a continuous periodic output

Phase-lock-loop (PLL)
A circuit which maintains a desired frequency by means of a self-correcting feedback technique

Plug-on
A transmitter type which may be attached directly to a microphone

PM
Phase modulation

Point-to-point
Transmission of signal from one fixed location to another fixed location

Polarization
The orientation of the electric field component of a radio wave

Power
Usually refers to the RF power delivered to the transmitter antenna, measured in milliwatts (mW)

Pre-emphasis
A fixed equalization which typically boosts high frequencies in the first part of a two-step noise reduction process

Propagation
Ability of radio signal to travel a significant distance

Protected channels
Designated TVBD-free TV channels at a given location, defined by FCC rules

PSK
Phase shift keying

QPSK
Quadrature Phase Shift Key, a digital transmission symbol coding with four possible values

Quadrature
Phase relationship of 90° between two signals

Quieting
The suppression of radio noise in a receiver when the desired signal is picked up at a certain strength

Radio waves
Electromagnetic waves which propagate a significant distance from the source

RCV
Abbreviation for Receiver

Receiver
Device which is sensitive to radio signals and recovers information from them

Reciprocal
Property of a passive antenna to operate equivalently either in transmit or in receive mode

Reflection
Retransmission of incident radio waves by metal objects

RF
Radio frequency, generally taken to mean well above 20,000 Hz

RFI
Radio frequency interference

Sampling
Conversion of a continuous audio waveform into discrete digital values

Selectivity
Measure of a receiver’s ability to discriminate between closely-spaced frequencies

Sensitivity
Measure of a receiver’s ability to respond to weak radio signals

Shadow
Blocking of radio waves by reflective or absorptive (lossy) objects

Shift keying
Modulation of a radio wave in discrete steps, used for digital signal transmission

Signal-to-noise ratio
Overall usable amplitude range of a signal or device, generally the difference between some reference level and the residual noise level

SINAD
A measure of receiver sensitivity stated as the RF signal strength required for given minimum signal-to-noise + distortion ratio

Spread spectrum
A radio transmission technique which spreads the energy of the signal over a wide frequency range rather than concentrating it at one frequency

Spurious emissions (spur)
Residual output from crystal-controlled transmitters occurring at frequencies that are offset from the operating frequency by multiples of the crystal base frequency

Spurious rejection
The ability of a receiver to reject spurious emissions

Squelch
Circuit in a receiver that mutes the audio output in the absence of the desired transmitter signal

Spectrum
A range of discrete frequencies

STL
Studio Transmitter Link, in the US: 944-952 MHz

Superheterodyne
In a receiver, the technique of filtering the received signal to eliminate possible image frequencies and then mixing the received signal frequency with the local oscillator (LO) to produce the intermediate frequency (IF)

Symbol
Digital data element that contains more than one bit, ie. a QPSK symbol contains two bits

Transmitter
Device which converts information to a radio signal

TVBD
Television Band Device, sometimes called White Space Device

TX
Abbreviation for transmitter

UHF
Ultra high frequency (about 300 - 3000 MHz)

VCA
Voltage controlled amplifier, used in audio compressor and expander circuits

VCO
Voltage controlled oscillator, used in radio tuning circuits

VHF
Very high frequency (about 30 - 300 MHz)

Wavelength
The physical distance between successive complete cycles of a wave, inversely proportional to frequency, dependent on properties of medium

White Space
Unoccupied space between active TV channels

Wideband
An FM signal in which the deviation is much greater than the modulating frequency
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BIOGRAPHY: Tim Year

Tim is a native of the south side of Chicago (Go White Sox!). A lifelong interest in both entertainment and science has led to the field of audio as his choice for combining these interests in a useful way. Prior to joining Shure he worked as an engineer for recording, radio and live sound, operated his own recording studio and sound company, and continues to play music professionally. He holds a BS degree in Aeronautical and Astronautical Engineering, with a minor in Electrical Engineering, from the University of Illinois, Urbana-Champaign. While at the University, Tim also worked as chief technician with both the Speech and Hearing Science and Linguistics departments.

Since joining Shure in 1984, Tim has served in a technical support and training capacity for multiple departments. He has been active in product and applications education for Shure customers, dealers, and installers, as well as company staff. His major goal has been to increase the understanding of quality audio by presenting technical information in a way that is thorough but still very accessible. Tim's particular emphasis is on the contribution of proper selection and technique for both wired and wireless microphones.

In addition, Tim has done technical presentations for many industry organizations (NAB, NAMM, AES, and SBE), as well as for US government entities such as the White House Communication Agency and the US Air Force. Through his international assignments he has been fortunate to be able to deliver presentations in more than twenty countries and on all but one of the continents (still waiting for an offer from Antarctica...).

He has provided specific applications assistance to various performing artists including the Rolling Stones and U2, for theme parks such as Disney and Universal Studios, and performance groups such as Cirque du Soleil.

While at Shure, Tim has authored several educational booklets including "Selection and Operation of Wireless Microphone Systems" and "Audio Systems Guide for Houses of Worship."

His articles have also appeared in Recording Engineer Producer, Live Sound Magazine, Pro AV, Technologies for Worship, and Church Sound Magazine.

I would like to cite the following individuals at Shure Inc. for their extreme patience and invaluable assistance:

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Our Dedication to Quality Products

Shure offers a complete line of microphones and wireless microphone systems for everyone from first-time users to professionals in the music industry— for nearly every possible application.

For over nine decades, the Shure name has been synonymous with quality audio. All Shure products are designed to provide consistent, high-quality performance under the most extreme real-life operating conditions.