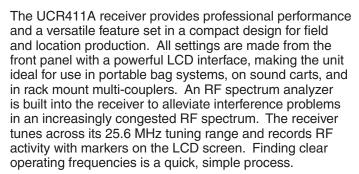
UCR411A

Compact Digital Hybrid Wireless Receiver

- Digital Hybrid Wireless™ Technology*
- SmartSquelch™
- SmartDiversity[™]
- SmartNR[™] noise reduction
- 256 selectable UHF frequencies
- Auto-tracking front-end filters
- DSP-based pilot tone
- Graphics type backlit LCD display
- Balanced XLR audio output
- Internal batteries or external DC powering options
- Compatibility modes for use with analog transmitters



The mechanical design of the receiver combines field proven features developed over many years of experience in motion picture and television production markets. The machined aluminum housing and panels are surfaced with electrostatic powder coated and anodized finishes with laser etched markings to withstand the rigors of field production.

Digital Hybrid Wireless™ technology* is a revolutionary new design that combines digital audio with an analog FM radio link to provide outstanding audio quality and the exemplary RF performance of the finest analog wireless systems.

The design overcomes channel noise in a dramatically new way, digitally encoding the audio in the transmitter and decoding it in the receiver, yet still sending the encoded information via an analog FM wireless link. This proprietary algorithm is not a digital implementation of an analog compandor. Instead, it is a technique which can be accomplished only in the digital domain.

The process eliminates a compandor and its artifacts, expanding the applications to include test and measurement of acoustic spaces and musical instruments.

*US Patent 7,225,135



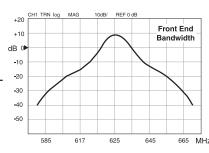
DSP Compatibility Modes

The native, compandor-free Digital Hybrid operating mode works with all Lectrosonics hybrid transmitters. DSP "compatibility modes" allow the receiver to also work with companded analog transmitters from Lectrosonics and some from other manufacturers. This expands the usefulness of the receiver and provides the backward compatibility needed to give existing customers an economical upgrade path.

Frequency Tracking Front-End Filters

The front-end consists of four transmission line resona-

tors with variable capacitance applied to each resonator to retune it as the frequency is changed. The tuning range covers a full 25.6 MHz block of frequencies. The design provides tunable, narrow filtering as selec-



tive as most fixed frequency designs, with the overload performance of the best front-ends available. The result is extended operating range in even the most congested RF environments.



SmartSquelch™

A conventional squelch design faces several compromises.

- Squelch too aggressively and audio may be lost.
- Squelch too little and excessive noise may be heard.
- Respond too rapidly and the audio will sound "choppy."
- Respond too sluggishly and entire words or syllables can be cut off.

SmartSquelch™ achieves an optimal balance of these tradeoffs by combining several techniques that remove distracting noise without the squelching action itself becoming a distraction. The circuitry will perform the following functions:

- · Wait for a complete word or syllable before squelching.
- Assess recent squelching history and RF signal strength.
- Assess audio content to determine available masking.

By adjusting squelching behavior dynamically for the optimal result under varying conditions, the receiver can deliver acceptable audio quality from otherwise unusable signals.

SmartDiversity[™]

Microprocessor controlled antenna phase combining is utilized for diversity reception to keep the receiver small, yet still deal effectively with multi-path dropouts. The embedded firmware analyzes RF level, the rate of change of RF level and the audio content to determine the optimum timing for phase switching, and the optimum antenna phase. This adaptive technique operates over a wide range of RF levels to anticipate dropouts before they occur. The system also employs "opportunistic switching" to analyze and then latch the phase in the best position during brief squelch activity.

SmartNR™

With a noise floor at -120 dBV and a frequency response to 20 kHz, high frequency noise in the source audio is more apparent than in conventional wireless systems. The Smart Noise Reduction algorithm works by attenuating only those portions of the audio signal that fit a statistical profile for randomness or "electronic hiss." Because it isn't simply a sophisticated variable low pass filter as in earlier analog designs, much greater transparency is obtained. Desired high frequency signals having some coherence such as speech sibilance and tones are not affected.

The Smart NR algorithm has three modes, selectable from the front panel LCD. When switched OFF, no noise reduction is performed. When switched to NORMAL, the factory default setting, enough noise reduction is applied to remove most of the hiss from the mic preamp and some of the hiss from lavaliere microphones. When switched to FULL, enough noise reduction is applied to remove most of the hiss from nearly any signal source of reasonable quality, assuming levels are set correctly at the transmitter.

Analog RF Links

A digitized audio or RF signal occupies a good deal more bandwidth than the original analog signal. A digital transmission over the air requires some combination of additional power, more RF bandwidth and/or compression of the audio data to achieve adequate operating range and keep the energy inside the defined spectral mask. Because of this, digital wireless microphones typically lack the operating range of conventional FM systems.

With regard to using RF power and spectrum efficiently, an analog RF link has many advantages in wireless mic systems, among them long battery life, excellent range, and the ability to use many systems in close proximity without interference.

DSP-Based Pilot Tone

The 400 Series system design utilizes a DSP generated ultrasonic pilot tone to control the receiver audio muting (squelch). Brief delays at turn-on and turn-off eliminate thumps, pops or other transients that can occur when the power is switched on or off. The pilot tone frequency is different for each of the 256 frequencies in the tuning range of a system (frequency block) to eliminate squelch problems in multichannel systems where a pilot tone signal can appear in the wrong receiver via intermodulation products. The DSP generated pilot tone also survives mishandling much better than fragile crystal-based pilot tone systems.

High Current, Low Noise Amplifiers

The gain stages in the front end use special transistors in a feedback regulated high current circuit that combines low noise, low gain, and high power. The design takes all three of these parameters into consideration at once, to provide low noise RF amplification, excellent sensitivity and extremely low susceptibility to intermodulation.

Combining the high power gain stages with the tracking front end produces a receiver that is immune to single and multiple interfering signals close to the operating frequency and in addition, strongly rejects signals that are much farther away.

Surface Acoustic Wave (SAW) Filter

SAW filters in the first IF section operating at 244 MHz combine sharp skirts, constant group delay, and wide bandwidth in one filter. These quartz filters are temperature stable. This special type of filter allows primary filtering as early as possible, at as high a frequency as possible and before high gain is applied to the signal. After the sharp filtering action of the SAW filters, the signal is converted to the second IF at 10.7 MHz, then finally to the third IF at the low frequency of 300 kHz, where the counting detector generates the audio signal.

Digital Pulse Counting Detector

An advanced digital pulse counting detector is used to demodulate the FM signal at 300 kHz to eliminate thermal drift and provide greater AM rejection. A stream of precision pulses is generated at 300 kHz and locked to the FM signal coming from the third IF section. The pulse width is constant but the timing between pulses varies with the frequency shift of the FM signal. The integrated voltage of the pulses in a given time interval within the waveform varies in direct proportion to the frequency modulation of the radio signal. Closely spaced pulses produces a higher voltage and widely spaced pulses a lower voltage. The resultant varying DC voltage is the audio signal, as shown in the diagram.

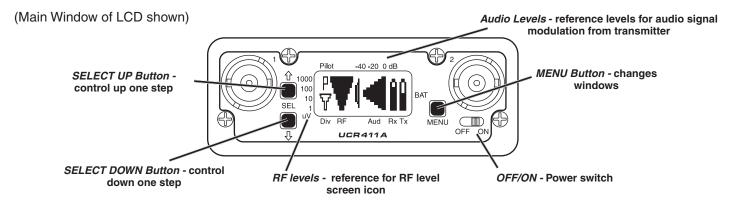
Timing between pulses varies with frequency **Digital Pulse Counting Detector**

Sum of pulse voltage increases at higher frequencies The varying DC voltage = The audio signal

Backlit LCD Graphics Display

The interface for setting up and operating the UCR411A receiver is an LCD display and three switches on the front panel of the receiver. All information regarding the status of audio and RF signals and battery levels in the receiver and the transmitter are simultaneously shown in the Main Window of the display.

A "locked" operating mode can be enabled by holding the MENU button down for several seconds. In this mode no changes can be made to the settings already stored, and the scanning mode cannot be entered.



Main Window

The LCD Main Window as shown above provides a summary of all receiver activity to display the status of the most important operating parameters at a glance. With quick button pushes, the LCD will also show frequency settings, battery status and setup functions.

The setup window provides control of the output level. audio test tone, transmitter battery type, output audio phase, noise reduction mode, tuning mode and compatibility mode.

Main Window Pilot Off/On Hold MENU & press UP SELECT Press & Hold MENU 4SETUP TV40 AE 631.800 EXIT **Setup Window** Frequency Window **Battery Level** Window Scan В8 Press All 3 Buttons Press Any Button

- Press Menu -

Fine View

B8

Scanning Windows

A special scanning mode is provided to conduct RF site surveys and locate clear operating frequencies.

A scrollable window provides a graphical illustration of RF signals (interference) within the tuning range of the receiver.

The receiver operates from external DC or internal 9 volt batteries. The battery door rotates around the XLR audio output jack, remaining attached to the receiver at all times. All parts are made of machined aluminum to endure the rigors of field production.

Protection against inserting the batteries backwards into the battery compartment is provided by both mechanical barriers and circuitry to prevent damage that might be caused by reverse polarity.

The housing is coated with an electrostatic powder coat finish and the battery door is anodized. Nomenclature is laser etched into the finishes for legibility and durability.



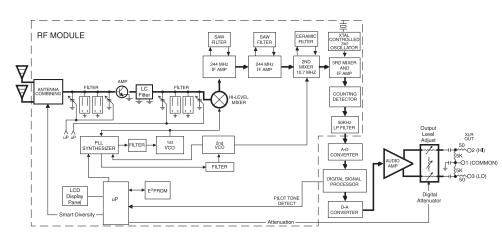
Operating Frequencies (MHz):

537.600 - 563.100

Block 22: 563.200 - 588.700 Block 23: 588.8 00-607.900 614.100 - 614.300 614.400 - 639.900 Block 24: Block 25: 640.000 - 665.500 Block 26: 665.600 - 691.100

Block 21:

Block 27: 691.200 - 716.700 Block 28: 716.800 - 742.300 Block 29: 742.400 - 767.900 Block 944: 944.100 - 951.900



Specifications

Frequency Adjustment Range: 25.5 MHz in 100 kHz steps Receiver Type: Triple conversion; superhet IF frequencies: 244 MHz, 10.7 MHz and 300 kHz

Frequency Stability: +0.001 % Front end bandwidth: ±5.5 MHz, @ -3 dB

Sensitivity: 20 dB Sinad: 0.9 uV

(-108 dBm), A weighted

60 dB Quieting: 1.12 uV

(-105 dBm), A weighted

AM rejection: >60 dB, 2 uV to 1 Volt

85 kHz Modulation acceptance: Image and spurious rejection: 85 dB Third order intercept: +8 dBm

Diversity method: Phased antenna SmartDiversity™

FM Detector: Digital pulse counting detector operating at 300 kHz Antenna inputs: Dual BNC female jacks; 50 Ohm impedance Audio output: Rear panel XLR connector; can drive 600 Ohm,

adjustable from -50 to +5 dBu in 1 dB steps

(into nominal 10 k bal. load)

Audio Performance (overall system):

(These audio performance specs apply only to 400 Series compatibility mode.)

32 Hz to 20 kHz (+/-1 dB) Frequency Response:

THD: 0.2% (typical) SNR at receiver output (dB):

Note: The dual envelope "soft" limiter provides exceptionally good handling of transients using variable attack and release time constants. Once activated,

SmartNR	No Limiting	w/Limiting
OFF	103.5	108.0
NORMAL	107.0	111.5
FULL	108.5	113.0

the limiter compresses 30+ dB of transmitter input range into 4.5 dB of receiver output range, thus reducing the measured figure for SNR without limiting by 4.5 dB

Input Dynamic Range: 125 dB (with full Tx limiting)

Overall Latency (time delay):

Audio Test Tone: 1 KHz, -50 to +5 dBu, <1%THD

Front Panel Controls and

Indicators: LCD display

• Menu - up/down buttons

Power switch

Rear Panel Features: · XLR audio output jack

• External DC input

· Battery compartment Minumum 10 Volts to maximum 18 VDC **External Power:**

1.6 W: 170 mA at 12 VDC

Battery Life: 6 to 8 hrs. continuous. w/ two alkaline 9 V batteries:

up to 20 hrs. continuous w/ two lithium 9 V batteries

Weight: 15.2 ozs. (with two alkaline 9V batteries) **Dimensions:** 3.23" wide x 1.23" high x 4.75" deep

