

The Way for 2-Way

# **Model 24-66 Best-in-Class VolP Base Remote Controller**

With our renewed commitment to VoIP / IP, Model 24-66 offers industry's best price performance remote control solution to meet diverse needs of our clients.

Being IP-centric it offers all the inherent advantages of IP connectivity. Model 24-66 VoIP Remote Controller offers multiple configuration options including Ethernet LAN, WAN and IP cloud based connectivity.

Model 24-66 VoIP Controller incorporates all the functionalities of our highly successful Tone and DC remotes.



Designed for use with the Model 24-66 VoIP desktop controller, the Model 20-28 VoIP interfaces to remotely located base stations or repeaters. Based upon proper control signals received from the desktop controller, the



Model 20-28 VoIP directly controls the radio by either binary or serial control.

# **Local Option**

For installations where local operation of the base station is desired, the 20-28 VoIP is available with desk microphone, speaker jack and volume control. (Option RTM-604)



# **Features & Capabilities**

- Handset or desk mic
- Up to 99 channels
- Intercom
- Mute
- Clock
- VU meter
- Alpha-numeric display
- 110 or 15 volt operation
- US or European power supply

#### **OPTIONAL FEATURES**

- One touch transmit
- Wall mount
- Headset and footswitch





















# **Model 24-66 VolP**

# **Specifications**



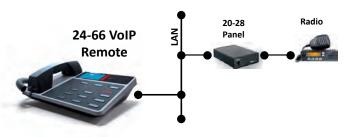
Input voltage	<b>TX Compression</b> With an audio increase of
24 Watt Wall Transformer	30dB beyond the start of compression
2.1 x 5.5mm barrel conn. Center (+)	the output increases 17.5dB
Current consumption @ 15Vdc450 mA (TX)	<b>Ethernet Interface</b> The Ethernet Interface consists of a
800 mA (RX)	three port switch. One port connects
400 mA (STBY)	externally to the LAN, the second port
Standby voltage	connects externally to a computer to allow
<b>Temperature range</b> 0 to +60° C	the computer access to the LAN, the third
Relative humidity90% at 50° C	port connects internally to the Ethernet
<b>TX hum &amp; noise</b> 60 dB (ref. +0 dBm)	port of the NET+ARM MCU.
Threshold of compression20 dBm adjustable	<b>DSP Firmware</b> The DSP firmware provides the
(speaker audio)	following standard CODEC algorithms:
Speaker audio output	G.711, G.723.1, G.726.
Distortion (at rated speaker output)< 3%	<b>Weight</b>
<b>RX hum &amp; noise</b> 47 dB (ref OdBm)	<b>Dimensions</b>
<b>Frequency response</b> +1, -3 dB (300 to 3000 Hz)	

#### **Overview**

The basic function of the voice over IP (VoIP) radio control system is to allow a private mobile radio base station to be controlled from a remote location. The system consists of a base station radio, a panel and a remote. The panel connects to the radio and provides direct control of the radio based upon control signals from the remote. The panel also has access to the radio's receive and transmit audio which is passed to and from the remote. The panel and the remote are connected by means of an IP network. This system will allow a user of the remote to change the current radio channel and select other radio functions in much the same way as can be done directly at the radio. The user of the remote can also carry on voice conversations with mobile radio users through the base station radio.

# **Basic Setup**

The basic VoIP radio control system is depicted in Figure 1. The remote and the panel are connected via an Ethernet LAN. If desired, the VoIP remote can also be connected to a computer which allows the computer access to the LAN without the need for a hub, thus eliminating the need for separate cabling.



**Basic Configuration** 

### **Extended Setup**

An extended VoIP radio control system is depicted in Figure 2. Here, multiple remotes connect to the panel across an internet. The internet might be composed of one or more Ethernet LANs, WANs or the Cloud.

#### 24-66 VoIP Remotes



