



Voice Functions and Features

- Protocol: SIP 2.0 (RFC 3261)
- Codec: G.711, G.723.1, G.726, G.728, G.729A/B, iLBC
- Built-in router, NAT and Gateway
- Powerful digital signal processing (DSP)
- Line Echo Cancellation (G.168), and AGC(Automatic Gain Control)
- Support NAT traversal using IETF STUN and symmetric RTP
- Voice activity detection (VAD) and comfort noise generation (CNG)
- Advanced adaptive jitter control and packet loss concealment

Calling Functions and Features

- Web, console, and telnet configuration
- Support Caller ID/name display or block
- Call waiting/Hold/Flash/Transfer/Forward
- 3-way conference
- In-band and out-of-band DTMF
- Support customizable ring tone (pending)

Network Functions and Features

- PPPoE, STUN, TFTP, DNS, DHCP (both client and server), NTP
- Support for Layer 2 (802.1Q VLAN, 802.1p) and Layer 3 QoS (ToS, DiffServ, MPLS)

Maintenance and Management

- AutoPro Remote Configuration Software for automatic programming of voip devices
- Support remote automated provisioning and software upgrade through firewall/NAT
- Support device configuration via built-in IVR, Web browser or central configuration file
- Support firmware upgrade via TFTP or HTTP with encrypted configuration files

Calling Functions and Features		
Supports SIP domain, SIP authentication DTMF Relay (In-band/Out-band & SIP Info)		
Port	WAN	1x10/100 Base-T RJ 45
	LAN	1x10/100 Base-T RJ 45 for PC
	1xRJ11 FXS 1xRJ11 Lifeline / Fallover	
NAT transverse	supports STUN client	
Operation Temperature	0~40℃ (32~104°F)	
Power Supply	Input: 100-240V	Output: 9V 1.2A
Dimension	240mm x 180mm x 100mm	
Weight	1KG	