



## GXP-2000 EXT Expansion Module



## GXP-2000 Enterprise IP Telephone



**GXP-2000EXT Expansion Module** features 56 fully programmable buttons (each with dual-color LED) when used with the GXP2000. The GXP-2000EXT is connected to and controlled by the GXP-2000 with a short connector cable. The GXP 2000 EXT creates up to 112 additional programmable extensions when 2 EXT are daisy chained with the GXP-2000. The GXP-2000 supports speed dialing, BLF (Busy Lamp Field), BLA (bridged line appearance) call transfer/forward/pickup on each of the programmable buttons on the GXP-2000EXT module.



**GXP-2000** is a next-generation enterprise SIP telephone that is feature rich, easy to use, supports Power-over-Ethernet and is competitively priced. It is ideal for small office and enterprise customers as well as service providers. The GXP-2000 is expandable, secure and easy to manage. It offers multiple lines, multiple SIP accounts, advanced functions, superior audio quality, and interoperability with most SIP end-user devices, IPPBX and Softswitches. The GXP-2000 received the prestigious "2006 Internet Telephony Excellence Award" and the "2005 Product of the Year" award from the Internet Telephony Magazine.

### Features

- **Open Standards Compatible** SIP RFC3261, TCP/IP/UDP, RTP/RTCP, HTTP, ARP, ICMP, DNS (A record and SRV), DHCP(both client and server), PPPoE, TFTP, NTP, TLS
- **Superb Audio Quality** Advanced Digital Signal Processing (DSP), silence suppression, VAD, CNG, AEC, AGC
- **Custom Ring-tone Software** Convert most music files to a custom ring-tone
- **Feature Rich** Standard voice features including caller ID, call waiting, hold, transfer, forward, Intercom, shared call appearances, mute, headset, autodial, off-hook dial, and click to dial
- **Advanced Functionality** Multi-line support, multi-party conferencing, line extensions, headset enabled, TLS & SRTP (pending), multi-language support (MLS) and XML enabled

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<b>Lines</b>	4 direct lines with 7 speed dial keys, up to 11 line calls (expandable to 112 lines with 2 daisy-chained GXP-2000 Ext)
<b>Protocol Support</b>	Support SIP RFC3261, TCP/UDP/IP, PPPoE, RTP/RTCP, SRTP, HTTP/HTTPS, ARP/RARP, ICMP, DNS, DHCP, NTP/SNTP, TFTP, TLS, SIMPLE/PRESENCE Support up to 4 SIP accounts
	Support for SIP PUBLISH method (RFC 3903), SIP Presence package (RFC 3856, 3863) for use of 7 MFks and GXP-2000EXT, SIP Dialog package (RFC 4235) Support SIP MESSAGE method (RFC 3428) and up to 100 incoming IM messages
<b>Display</b>	8 line x 22 character, 64 rows x 130 columns in pixels
<b>Feature Keys</b>	8 dedicated keys: Message Button, Hold, Transfer, Conference, Speakerphone, Send, Mute/Del 5 display/menu navigation keys, dual color LEDS
<b>Telephony Features</b>	Intuitive graphic user interface, downloadable phone book (XML, LDAP), multi-language support(MLS) Support customizable LCD screen via downloadable XML by HTTP/TFTP Dynamic negotiation of codec and voice payload length G.723,1 (6.3K/5.4K), G.729A/B, G.711 $\mu$ /A, G.726 (40K/32K/24K), G.722 (wide-band), GSM and iLBC Call hold, call transfer (attended/blind), do-not-disturb, call forwarding, mute Redial, call log, and volume control Caller ID display or block Call waiting, call waiting caller ID 3-way conferencing Voice mail indicator Downloadable custom ring tones In-band and out-of-band DTMF (in audio, RFC2833, SIP INFO) Silence suppression, VAD, CNG, side tone Full duplex hands-free speakerphone with Acoustic Echo Cancellation (AEC) and AGC Off-Hook Auto Dial, Auto Answer, Early Dial and Speed Dial Adaptive jitter buffer control with packet delay and loss concealment

<b>Network and Provisioning</b>	Via keypad/LCD, Web browser, or secure (AES encrypted) central configuration file Support NAT traversal using STUN and symmetric RTP Support static IP and DHCP Support Layer 2 (IEEE 802.1p/Q VLAN tagging) and Layer 3 QoS (ToS, DiffServ)
<b>Firmware Upgrades</b>	Support auto-provisioning via secure configuration file with authentication verification Support NAT-friendly firmware upgrade via TFTP or HTTP Allow user to specify different URL for configuration file and firmware files
<b>Advanced Server Features</b>	Support DNS SRV look-up and SIP server fail-over
<b>Security</b>	DIGEST authentication using MD5/MD5-sess, SIP over TLS and SRTP (pending) secure configuration file (AES)

### Hardware Specification

Hardware	Specifications
<b>Network Interfaces</b>	Two RJ45 10/100 Mbps auto-sensing Auto detect Ethernet polarity, allowing straight-through or twisted cable
<b>Power over Ethernet</b>	Integrated (IEEE 802.3af)
<b>Headset Jack</b>	2.5mm Headset
<b>LED</b>	11 LEDs with different light patterns
<b>Universal Switching Power Supply</b>	Input: 100-240VAC 50-60 Hz Output: +5VDC, 1.2A, UL certified
<b>Dimension (LxWxH)</b>	220mm x 215mm x 57mm
<b>Weight</b>	0.82kg (1.8lbs)
<b>Temperature</b>	32 - 104°F / 0 - 40°C
<b>Humidity</b>	10% - 90% (non-condensing)
<b>Compliance</b>	FCC / CE / C -Tick

