



GXW4004/4008

4/8 port FXS Gateways

The GXW400x FXS gateway series enables small and medium businesses to create integrated IP and PSTN telephony systems that efficiently manage communication costs. These analog FXS gateways feature 4 or 8 FXS ports for analog telephones, two 10/100 Mbps ports for connection to an IP network. The GXW400x also includes a built-in NAT router. Advanced telephony features, easy automated provisioning superb voice quality and a PSTN lifeline port allow the GXW400x series to be the ideal analog gateway solution for businesses.



TLS and SRTP security encryption technology to protect calls and accounts



Automated provisioning options include TR-069 and XML config files



3 WAY
Supports 3-way voice conferencing



Failover SIP server feature automatically switches to secondary server if main server loses connection



Supports T.38 Fax for creating Fax-over-IP



Supports a wide range of caller ID formats

**zero
CONFIG**

Use with Grandstream's UCM series of IP PBXs for Zero Configuration provisioning



Supports advanced telephony features, including call transfer, call forward, call-waiting, do not disturb, message waiting indication, multi-language prompts, flexible dial plan and more

Telephone Interfaces	GXW4004: 4RJ11 FXS ports GXW4008: 8 RJ11 FXS ports
Network Interfaces	10M/100Mbps, dual RJ45 ports
Admin Console Interface	RS232 Serial Port
LED indicators	Power, Network and Line LEDs
Voice-over-Packet Capabilities	G.168 compliant Echo Cancellation, Dynamic Jitter Buffer, Modem detection & auto-switch G.711
PTSN Fail-over Life Line	PTSN failover on power failure
Voice Compression	G.711, G.723, G.726 (40/32/24/16), G.729A/B/E, G.728, iLBC
DHCP Client/Server	Yes, NAT Router or Switched Mode
Fax over IP	T.38 compliant Group 3 Fax Relay up to 14.4kbps and auto-switch to G.711 for Fax Pass-through
QoS	Diffserve, TOS, 802.1P/Q VLAN tagging
IP Transport	RTP/RTCP
DTMF Method	Flexible DTMF transmission method, user interface of in-audio, RFC2833, and/or SIP info
IP Signaling	SIP (RFC 3261)
SIP Server Profiles & Accounts Per System	Up to 2 distinct SIP server profiles per system and independent SIP account per port
Provisioning	TFTP, HTTP, HTTPS
Media	SRTP not available
Control	TLS/SIPS
Management	Syslog support, HTTPS, telnet, remote management using web browser
Power	Output: 12VDC Input: 100-240VAC / 50-60Hz
Environmental	Operational: 32-104°F or 0-40°C Storage: 14-140°F or -10-60°C Humidity: 10-90% Non-condensing
Dimensions (LxWxH)	230mm x 135mm x 35mm
Mounting	Wall mount and desktop
Short & Long Haul	REN3: Up to 150ft on 24AWG line
Caller ID	Bellcore Type 1&2, ETSI, BT, NTT, and DTMF-based CID
Polarity Reversal/Wink	Yes
EMC	EN55022/EN5504 and FCC part 15 Class B
Safety	UL