

Turning Copper into Gold GST No. 07AAAECA9973H1ZA

ATX4SA/ATX8OA VOIP Developed for small businesses, SOHO and residential IP telephony applications, the ATX4SA/ATX8OA gateways are integrated VoIP devices that provide reliable and cost efficient access methods flexible for deploying converged communication solutions for network operators and enterprises as well. ATX4SA/ATX8OA are designed as compact units with 2 or 8 FXS/FXO ports, 1 WAN port, and 1 PC port, which allow users to apply them as N-to-1 systems capable of providing versatile solutions by connecting analog phone, fax and POS machine, IP telephony and PSTN. Comprehensive Feature Set with Intelligent Engine As intelligent gateways running on embedded Linux operating system, ATX4SA/ATX8OA are compatible with many highlighted features such as call forward, call transfer, 3-way conference, caller ID, DND, color ringback, forking, automatic routing, digitmap, PSTN failover, and much more to provide a complete and highly reliable VoIP solution applicable to most scenarios. High Reliability

The hardware design of the ATX4SA/ATX8OA, from architecture to component selection, take full consideration of improving their reliability. In addition, with high availability featuring primary-standby, active-standby and load balance as well as PSTN failover designed to handle the failure of the SIP proxy servers, power supply and network, the users are provided with carrier class reliability with minimized potential downtime. Robust Protection against Intrusions ATX4SA/ATX8OA provide rich set of features to support perimeter protection against malicious intrusions, such as brute force logins and SIP attacks, ensuring a high level of security in your IP telephony environment. Remote Management and NAT Traversal The ATX4SA/ATX8OA include the Cloud client, providing easy access with NAT traversal to endpoint devices and value-added applications for remote, real-time monitoring, maintenance, software upgrades, troubleshooting, data analysis, alarm notification, remote and packet capture

Features

Flexible configuration of FXS/FXO ports
SIP/MGCP protocols
500 routing and number manipulation rules
Support Smart Fax over IP (SFoIP)
PSTN failover
TLS/SRTP
Auto provisioning
Remote access via Cloud (US: YoubiquITy)
Management with or third-party EMS (TR-069, SNMP)
Interoperability with popular SIP servers, such as Cisco
CallManager/
CUCM, Broadsoft, Microsoft Skype for Business (Lync),
Huawei IMS,
VoS, FreeSWITCH and Asterisk/Elastix
Class I lightning protection





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Specification

	SIP/UDP and SIP/TCP (RFC3261),	
Call control	IMS (3GPP), MGCP (RFC3435)	
Can control	1.1.2 (2.311), 1.1.3.21 (1.1.2.2.1.2.1)	
	SSH, HTTP, HTTPS, DHCP/PPPoE	
Network	client,	
	Radius, DNS (A/SRV record), STUN	
Media	/ \ //	
Processing		
Trocessing	D II /FI II FEGI DE DELCE	
	Bellcore/Telcordia, ETSI, BT, DTMF	
	G.711 (a/µ), G.729a, G.722	
Caller ID	In-band audio, RFC2833, SIP-INFO	
Codec	Local processing, RFC2833, SIP-INFO	
DTMF	T.38, G.711 pass-through	
Hook-flash	T.38 compliant Group 3 Fax Relay	
Fax over IP	Maximum fax rate of 33,600 bps (T.38)	
	Support	
	Smart Fax over IP (SFoIP)	

Disconnect modes Polarity reversal, Busy tone detection, Loop current

QoS: DiffServ, TOS, 802.1P/Q VLAN tagging

Voice-quality:enhancement Echo cancellation (G.168-2004), Jitter buffer, Silence suppression (VAD, CNG), PLC

HTTPS, SSH, TLS, SRTP Encrypted password/PIN

Brute force login protection, Access list, Firewall, SIP-allowed IP addresses, Alarm and notification* Auto discovery via DHCP, Manual configuration L2TP. OpenVPN

Blind transfer, Consultative transfer

Call transfer: Call forward all, Call forward no answer,

Call forward: Call forward busy

Call recording storage Remote recording server for storage Call settings Routing based on the caller or callee number, Digit map, 3-way calling, Speed dialing, Do not disturb, Call forking, Color ringback tones Hunt group, Ring cadence, Message Waiting Indicator (MWI) EMS, TR-069 management (TR-069, TR-104 and TR-106), SNMP

Protocol HTTPS, SSH, TLS, SRTP **Encryption**: Encrypted password/PIN

Intrusion prevention: Brute force login protection, Access list, Firewall, SIP-allowed IP addresses, Alarm and notification* VLAN: Auto discovery via DHCP, Manual configuration

VPN client: L2TP, OpenVPN

Redundancy	Primary-Standby, Active-Standby, Load balancing	
PSTN failover	Power failure, Network failure	

Device management

Auto provisioning Download configuration file via TFTP/FTP/HTTP/ HTTPS, Obtaining ACS address via DHCP option 66 or redirection

Log management: 8-level logs, Syslog **Data capture**: Port capture, Packet capture

Status and statistic: Call status and history, Device status

monitoring and statistics collection

Upgrade : Firmware upgrade via Web GUI or Auto provisioning

Included Software

Finder Obtains IP addresses of devices in a LAN environment

Telegreeting: Formats pre-prepared IVR audio files for uploads them into a device

Recording Agent :Receives recording streams in PCM and converts them into MP3 format. Includes management functions such as search, play, delete and backup

Hardware:

CPU	700 MHz	
RAM	64 MB	128 MB
Flash	16 MB	32 MB
H×W×D (1U)	30 × 150 × 109 mm	35 × 228 × 135.5 mm
Weight (net)	0.3 kg	0.825 kg
Power	12VDC/1.5V	12VDC/1.5V
Power consumption	6 W	12 W
Mounting	Desktop	
Operating	Temperature: 0 to 40°C Humidity: 10 to 90% RH (non-condensing)	
Storage	Temperature: -40 to 70°C Humidity: 5 to 90% RH (non-condensing)	

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