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Features

SIP/MGCP protocols

500 routing and number manipulation rules

Fax over IP using T.38 & G.711 pass-through

Dual redundant Ethernet ports

Dual DC/AC power supplies available

PSTN failover

TLS/SRTP

Auto provisioning

Remote access

Management or third-party EMS (TR-069, SNMP) Interoperability with popular SIP servers, such as Cisco Call Manager/

CUCM, Broadsoft, Microsoft Skype for Business (Lync), Huawei IMS,

VoS, Free SWITCH and Asterisk/Elastix Class I lightning protection

Specification

Call control	SIP/UDP and SIP/TCP (RFC3261), IMS (3GPP), MGCP (RFC3435) SSH, HTTP, HTTPS, DHCP/PPPoE client, Radius, DNS (A/SRV record), STUN			
Network		·		
Caller ID Codec	Bellcore/Telcordia, ETSI, BT, DTMF G.711 (a/μ), G.729a, G.723.1, GSM, iLBC			
Fax over IP				
Disconnect modes				
Voice-quality enhancement Call transfer Call forward		Echo cancellation (G.168-2004), Jitter buffer, Silence suppression (VAD, CNG), PLC		
Call recording				
storage				
Call settings Blind transfer, Consultative				
transfer				
Call forward all, Call forward no answer, Call				

The AT-60XE is a reliable, cost-effective, high-performance VoIP gateway

with up to 48 FXS/FXO ports, designed to meet the needs of high-traffic

contact center and enterprise applications.

The feature-rich AT-60XE allows users to consolidate analog phones, fax

machines, and POS terminals, configurable with up-to-full concurrency over

SIP trunks and the PSTN.

Comprehensive Feature Set

As an intelligent gateway running on embedded Linux operating system,

the AT-60XE supports an advanced feature set such as call forward, call

transfer, 3-way calling, caller ID, DND, color ringback tones, call forking,

automatic routing, digit map, and much more, providing a complete and

highly reliable VoIP solution.

Carrier-class Reliability

The AT-60XE provides high-availability configurations with SIP registration

failover, load balancing and PSTN failover with optional 1+1 redundancy

of Ethernet ports and power supplies, virtually eliminating downtime.

Robust Protection against Intrusions

AT-60XE provides a 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 AT-60XE includes the Cloud client, providing easy access

with NAT traversal to endpoint devices and value-added applications for

remote, real-time monitoring, maintenance, software upgrades,

ANDA Telecom Pvt. Ltd.

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forward busy	
Remote recording server for	
storage	
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)	

VPN client* L2TP, OpenVPN

VLAN Auto discovery via DHCP, Manual configuration

Intrusion prevention

Brute force login protection, Access list, Firewall, SIP-allowed IP addresses, Alarm and notification*

Encryption Encrypted password/PIN

Protocol

T.38, G.711 pass-through T.38 compliant Group 3 Fax Relay Maximum fax rate of 33,600 bps (pass-through)
Polarity reversal, Busy tone detection, Loop current
HTTPS, SSH, TLS, SRTP

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

troubleshooting, data analysis, alarm notification, remote and packet capture.

Hardware

Hardware		
CPU	1GHz	
Voice DSP module	200 MHz, 2 to 6 modules	
RAM	128 MB, DDR3	
Flash	32 MB	
H×W×D (1U)	44 × 440 × 260 mm	
Weight (net)	3 kg maximum	
Single/Dual AC power supplies	100 to 240V AC, 50/60 Hz, 1.2 A maximum	
Power consumption	60 W	
Mounting	Rack	
Single/Dual DC power supplies	-36 to -72V DC, 2A	
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)	

Receives recording streams in PCM and converts them into MP3 format. Includes management functions such as search, play, delete and backup DTMF In-band audio, RFC2833, SIP-INFO

Redundancy PSTN failover	Primary-Standby, Active- Standby, Load balancing Power failure, Network failure
Device management Remote access over TCP	Cloud (US: YoubiquITy) EMS, TR-069 management (TR-069, TR-104 and TR-106), SNMP

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