

Analog VoIP Gateway (MediaPack™ 5xx)

The AudioCodes **MediaPack™ 5xx series of analog VoIP gateways** provide service providers and enterprises with superior voice technology for connecting legacy telephones, fax machines and PBX systems with IP telephony networks, IP-PBX systems and unified communications solutions.



MediaPack 5xx gateways offer comprehensive SIP interoperability with leading vendors, delivering secure and reliable communications for enterprises and branch office locations.

4 to 32 Analog ports | Zero-Touch Provisioning | Branch Survivability | T.38 Fax Compliant



Comprehensive interoperability

Proven interoperability with SIP trunks, SIP platforms and IP cloud services



High resiliency

Standalone survivability (SAS) and fallback to PSTN for E911 (emergency number PSTN breakthrough) or upon network/power failure



Superior voice quality

Advanced capabilities for optimizing and monitoring voice service quality



Enhanced capabilities

Includes MWI, long haul, metering tones generation and caller ID



All-in-one option

Option to add data routing capabilities within the same device

Interfaces		
	MP-504 and MP-508	MP-516, MP-524, MP-532
Telephony Interfaces	2/4/8 ports, FXS loop-start	16/24/32 ports, FXS loop-start D-sub or RJ11 assembly options
Network Interfaces	1x WAN copper Ethernet 10/100/1000; 1x LAN 10/100/1000Base-TX	2x WAN copper Ethernet 10/100/1000; 1x LAN 10/100/1000Base-TX
Voice Features		
Voice Coders	G.711, G.723.1, G.729A, G.722	
VoIP Features	G.165 and G.2002-168, with 64 ,32 or 128 msec tail length, silence detection and comfort noise generation, dynamic programmable jitter buffer	
Call Progress Tones	Programmable detection and generation tones, pre-recorded tones generation	
DTMF	In-band and out-of-band signaling, RFC 2833, SIP INFO/NOTIFY	
Fax Transport	T.38 compliant (real time fax), automatic bypass to PCM	
Voice Security - Session Border Controller (SBC)	<ul style="list-style-type: none"> • SIP header conversion, SIP normalization • Survivability • IP-to-IP routing translations of various SIP transport types; UDP, TCP, TLS, translation of RTP, SRTP without reduction • Support SIP trunk with multi-ITSP (registrations to ITSPs are invoked independently) • Topology hiding • Call Admission Control • Call black/white list 	
Subscriber Line Interface		
3-Way Conference	On-board 3-way conference	
MWI	Message Waiting Indication: High (neon) and low (LED) voltage, FSK, stutter dial tone	
Caller ID	Bellcore GR-30-CORE Type 1 using Bell 202 FSK modulation, ETSI Type 1, NTT, Denmark, India, Brazil, United Kingdom and DTMF ETSI CID (ETS 300-659-1)	
Polarity Reversal/Wink	Immediate or smooth to prevent erroneous ringing	
Metering Tones	12/16 KHz sinusoidal bursts, generation on FXS	
Distinctive Ringing	By frequency (15-100 Hz) and cadence patterns	
Analog Telephony Features		
Ring Voltage	Sine: 54 VRMS typical (balanced ringing only)	
Ring Frequency	15-50Hz	
Maximum Ringer Load	REN3	
Loop Impedance	Up to 800 ohm (including phone impedance)	
Line Current	Up to 25 mA on 4 ports	
Control and Management		
Control Protocols	<ul style="list-style-type: none"> • SIP-TCP, SIP-UDP, SIP-TLS and IPv6 supported 	
Operations & Management	<ul style="list-style-type: none"> • AudioCodes' One Voice Operation Center • Embedded HTTP Web Server, SNMP V2/V3, SSH, Telnet, TR-69, TR-098, TR-181, TR-104 • User authentication and access control via HTTP or HTTPS, RADIUS, TACACS, Syslog (for events and alarms) • Zero-touch provisioning 	
Optional Data Routing and Security		
Routing Functionality	<ul style="list-style-type: none"> • IPv6, IPv6/IPv4 Dual Stack, ICMPv6, DHCPv6, SLAAC • VLAN and IEEE 802.1Q VLAN tagging • DHCP client, relay, server • Layer 3 routing and layer 2 bridging, jumbo frames • Static and dynamic routing (RIP1, RIP2, OSPFv2, BGP), policy-based routing 	
Data Security	<ul style="list-style-type: none"> • Stateful packet inspection firewall • Network Access Control - 802.1x • IPSec with IKEv2 • ESP – Tunnel mode • Encryption protocols: 3DES, AES, SHA-256, MD-5 	<ul style="list-style-type: none"> • IDS/IPS (fragmented traffic, malformed request, ping of death, properly formed request from unauthenticated source, DDoS attack, SYN flood) • Packet filtering
Quality of Service	<ul style="list-style-type: none"> • IEEE 802.1P, DSCP, TOS, DiffServ labeling, WRED, • Marking, policing, and shaping, class-based queuing with prioritization, queuing based on VLAN 	
Physical/Environmental		
	MP-504 and MP-508	MP-516, MP-524, MP-532
Power	100-240V, 50-60 Hz, 0.8A	90-240,50-60Hz, 1.5A
Environmental	Operational: 5 to 40°C (41 to 104°F) Storage: -25 to 85°C (-13 to 185°F) Humidity: 10 to 90% non-condensing	
Dimensions (HxWxD)	51 x 296 x 160 mm (HxWxD)	19"(443mm)x1Ux288mm
Mounting	Table top, wall mount, rack mount (with an optional shelf)	
Weight	1.5 lb (670g)	5.16 lb (2.34Kg)

* For specific configuration availability please contact an AudioCodes sales representative

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