

AudioCodes CPE & Access Gateway Products

Mediant™ 1000 VoIP Media Gateway



- Employs AudioCodes VoIPerfect™ technology for outstanding voice quality
- Scalable “pay-as-you-grow” modular architecture
- Rich offering of digital (E1/T1/J1) and analog (FXS/FXO) interfaces
- Cost-efficient for low density gateways
- Life-line fallback to PSTN in case of power failure or network degradation
- PSTN fallback for assured connectivity
- Ideal platform for OEM VoIP applications

The **Mediant™ 1000** is AudioCodes' cost-effective, converged wireline and wireless VoIP media gateway utilizing cutting edge technology. Intelligently packaged in a stackable 1U chassis, it is designed to interface between TDM & IP networks in enterprises or small-scale carrier locations. Incorporating AudioCodes' innovative Voice over Packet technology, the Mediant 1000 enables rapid time-to-market and reliable cost-effective deployment of next-generation networks.

The Mediant 1000 is based on VoIPerfect™, AudioCodes underlying, best-of-breed, media gateway core technology for all of its products. The Mediant 1000 provides superior voice technology for connecting legacy telephone and PBX systems to IP networks, as well as seamless connection of the IP-PBX to the PSTN. In addition to operating as a pure media gateway, the Mediant 1000 can also host partner applications and serve as an IP-PBX platform. The Mediant 1000 is fully interoperable with multiple vendor gateways, softswitches, gatekeepers, proxy servers, IP phones, session border controllers and firewalls.

SCALE UP AS YOUR BUSINESS GROWS

The Mediant 1000 matches the density requirements for smaller locations while meeting service providers' demands for scalability. The compact Mediant 1000 Modular Gateway is extremely scalable and supports multiples of 1, 2, or 4 E1/T1/J1 spans, or 1 to 24 analog ports in various FXO/FXS configurations. The Mediant 1000 will also support mixed digital/analog configurations.

The Mediant 1000 can support a variety of telephony interfaces. It contains up to 4 digital or up to 6 analog interface modules with 4 ports. The digital module can be configured as regular E1/T1/J1 interfaces, with up to 1 or 2 paired spans acting as life-line interfaces for switching to the PSTN in case of power failure or network problems. The analog module is available as regular FXS or FXO interfaces, where 1 FXS line can be used as a life-line interface for switching to the PSTN.

- Digital – connecting the PSTN or PBX to the IP-network
- Analog FXS – connecting analog phones and fax machines to the IP-network
- Analog FXO – connecting analog lines from the Central Office (CO) or PBX to the IP network

SEAMLESS INTERFACE WITH LEGACY ENTERPRISE NETWORKS

The Mediant 1000 has enhanced hardware and software capabilities to ease its installation and to help maintain voice quality. If the measured voice quality falls beneath a pre-configured value, or the path to the destination is disconnected, the Mediant 1000 can assure voice connectivity by falling back to the PSTN. In the event of network problems, calls can be routed back to the PSTN without requiring routing modifications in the PBX. Further reliability is provided by dual Ethernet ports and optional dual AC power supply.

FLEXIBLE PLATFORM FOR PARTNER APPLICATIONS

The Mediant 1000 extends the flexibility of the Media Gateway family with additional deployment options. The open platform on the Mediant 1000 offers partners the option to host their own applications (e.g., IP-PBX or call center application) using a powerful, low-power processor and hard disks to provide a complete solution within the Mediant 1000 chassis.

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SPECIFICATIONS

Interfaces	
Modularity and Capacity	6 slots for analog or 4 slots for digital modules Up to maximum of 24 analog ports or 4 digital spans
Digital Modules	1,2 or 4 E1/T1/J1 spans using RJ-45 connectors per module Up to 4 digital modules (maximum 4 spans per gateway) Optional 1+1 or 2+2 fallback spans
Analog FXO and FXS Modules	2 or 4 ports using RJ-11 connectors per module; up to 6 modules per gateway One life-line port per FXS module (in case of power failure or network problems)
I/O	MOH (Music On Hold), NB (Night Bell)
Ethernet	Dual Redundant 10/100 Base-TX Ethernet ports via 2 RJ-45 connectors on RTM
RS-232	For basic configuration and debugging
Media Processing	
Voice Coders	G.711, G.726, G.723.1, G.729A, EVRC, AMR ³ and GSM-FR Independent dynamic vocoder selection per channel
Echo Cancellation	G.165 and G.168 2000, with 32, 64 or 128 tail length ²
QoS Features	Dynamic programmable jitter buffer, VAD, CNG, 802.1p/q VLAN tagging, DiffServ, voice quality monitoring
DTMF/MF Transport	Packet side or PSTN side detection and generation, RFC 2833 compliant DTMF relay Call Progress tones Detection and Generation
IP Transport	VoIP (RTP/RTCP) per IETF RFC 3550 and 3551
Fax and Modem Transport	T.38 compliant (real time fax), Automatic bypass to PCM or ADPCM
Partner Application Platform	
Single Chassis Integration	Embedded, Celeron™ based platform for third party hosted applications
CPU	Intel™ Celeron™ 600 MHz
Memory	Two SODIMM slots
Storage	Single/Dual hard disk drives
Interfaces	10/100 Base-TX, USB, RS-232, NB relay, MOH
Signaling	
Digital –PSTN Protocols	CAS: MF-R1: Wink Start, delay dial, immediate start, FGB, FGD and MFC/R2: numerous country variants with unique script for each variant. CCS: ISDN PRI: ETSI/EURO ISDN, ANSI N12, and other variants (DMS, 5ESS)
Analog Signaling	FXS, FXO Loop-start ; Caller ID; polarity reversal; metering tones, distinctive ringing, visual message waiting indication
Control & Management	
Control Protocols	SIP, H.323, (MEGACO – for digital trunks only)
Operations & Management	AudioCodes Element Management System Embedded HTTP Web Server, Telnet, SMDI Remote configuration and software download via TFTP, HTTP, HTTPS, DHCP and BootP, RADIUS, Syslog (for events, alarms and CDRs)
Security	
	IPSEC, HTTPS, TLS, SSL, Web access list
Hardware Specifications	
Power Supply	Single universal 90-260 V AC or dual load sharing, redundant power supply
Physical	1U high, 19-inch wide rack mount
Regulatory Compliance	
Telecommunication Standards	TIA/EIA-IS-968, TBR-4, TBR-13, and TBR-21
Safety and EMC Standards	UL60950-1; 47 CFR part 15 Class A, CE Mark (EN55022 Class A, EN60950-1, EN55024, EN300 386)
Environmental Specifications	ETS 300019-2-1 Storage T1.2, ETS 300019-2-2 Transportation T2.3 ETS 300019-2-3 Operating T3.2

APPLICATIONS

- PBX Networking
- IP-Centrex/Hosted IP-PBX
- Partner Applications (e.g., IP-PBX, Call Center)
- Remote Office Applications

ABOUT AUDIOCODES

AudioCodes Ltd. (NASDAQ: AUDC) enables the new voice infrastructure by providing innovative, reliable and cost-effective Voice over Packet technology and Voice Network products to OEMs, network equipment providers and system integrators. AudioCodes provides its customers and partners with a diverse range of flexible, comprehensive media gateway and media processing technologies, based on VoIPerfect™ – AudioCodes' underlying, best-of-breed, core media gateway architecture. The company is a market leader in voice compression technology and is a key originator of the ITU G.723.1 standard for the emerging Voice over IP market. AudioCodes voice network products feature media gateway and media server platforms for packet-based applications in the wireline, wireless, broadband access, and enhanced voice services markets. AudioCodes enabling technology products include VoIP and CTI communication boards, VoIP media gateway processors and modules, and CPE devices.

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¹ Ask your AudioCodes sales rep or distributor for availability of mixed analog and digital trunks
² 128 msec may cause capacity reduction
³ Contact AudioCodes