

Mediatrix® G7 Series

The Mediatrix G7 Series is a reliable and secure Media Gateway with VoIP Analog Adaptor capabilities for SMBs. Featuring E1/T1 PRI, ISDN BRI, FXS, and FXO interfaces; the Mediatrix G7 Series provides the best solution to connect legacy equipment to cloud telephony services and IP PBX systems to PSTN landlines.

Widely interoperable with SIP softswitch and IMS vendors, the Mediatrix G7 Series enable transparent integration of legacy PBX systems for SIP Trunking and PSTN replacement applications.



Interconnects any device to SIP

The Mediatrix G7 Series links any analog or digital connection to an IP network and delivers a rich feature set for a comprehensive VoIP solution.

PSTN access and legacy PBX system gateway

With configurable E1/T1 PRI, ISDN BRI, FXS, and FXO ports, local call switching, and user-defined call properties (including caller/calling ID), Mediatrix G7 Series gateways smoothly integrate into legacy PBXs and incumbent PSTN networks.

Highly reliable Fax and Modem Transmissions over IP

With T.38 and clear channel fax and modem pass-through capabilities, the Mediatrix G7 Series ensures seamless transport of voice and data services over IP networks.

Advanced Mass Management

Our advanced provisioning capabilities deliver remarkable benefits to Mediatrix customers. Mediatrix enables centralised CPE management, a definite advantage to monitor the network, ensure service, and reduce operational costs.

Mediatrix® G7 Series

Applications

Operators

- ✓ Connect legacy equipment in PSTN replacement/TDM replacement projects
- ✓ Provide SIP termination for cloud telephony services
- ✓ Convert ISDN signaling to SIP for SIP trunking
- ✓ Convert analog signaling to SIP for Hosted Unified Communications and IP-Centrex

System Integrators

- ✓ Integrate Unified Communications with legacy systems
- ✓ Connect Skype for Business with IP/TDM trunks and legacy telephony equipment
- √ Keep existing telephony equipment in SIP migrations
- ✓ Inter-connect branch offices to headquarters
- ✓ Survivability for branch offices in case of WAN failure

Key Features

Carrier-Grade Voice Quality

T.38 and clear channel fax over IP
High performance processing of up to 120 voice channels
Survivability for IP-Phones in Hosted UC/PBX
deployments
Battery reversal for pay phones

Robust Security

Encrypted media, signaling, and management

Deep packet inspection firewall with DoS protection

Easy Configuration and Management

Zero-touch configuration Intuitive Web GUI Customisable factory settings

Networking

Dual-stack IPv6 and IPv4
Multiple IP addresses and VLANs
NAT, firewall, and router capabilities

Benefits

- ✓ High quality built and carrier-grade validation standards contribute to the industry's most reliable platform.
- ✓ Extensive TR-069 support for an easy management of large-scale deployments with a centralised EMS.
- ✓ Superior call routing and manipulation allow greater flexibility in the implementation of complex deployment scenarios.

Technical Specifications

Media Processing

G.711 (A-law and μ -law), G.722, G.726, and G.729a/b; G.168 echo cancellation DTMF detection and generation Carrier tone detection and generation Silence detection/suppression and comfort noise Configurable de-jitter buffer and packet length Packet loss concealment

Enhanced Security

Denial of Service (DoS) protection SIP over TLS SRTP with AES cipher - 128 bits MIKEY key management protocol (RFC 3830 and 4567) SDES key management protocol (RFC 4568) TLS-encrypted configuration and management X.509 certificate management OCSP (Online Certificate Status Protocol) revocation status verification TLS Version 1.2 Secure TLS ciphers like ECDHE with AES-256 and SHA-

Management

Zero-touch provisioning TR-069, TR-104, and TR-111 Web GUI SSH and TELNET SMNP v1, v2c, and v3 Scripts/firmware files uploaded via HTTP, HTTPS, FTP, and TFTP Dual firmware banks Multiple levels of management access rights Customisable CDR Event notifications via Syslog, SIP, log file, and SNMP traps Remote activation of service licenses

Monitoring and Troubleshooting

Alarms and traps Call quality reporting (eMOS) (RTCP-XR as per RFC 6035) Call Details Record (CDR) Media quality statistics System: CPU and memory usage PCM capture IP network capture Diagnostic traces

Quality of Service (QoS)

Bandwidth limitation and traffic shaping TOS/DiffServ IEEE 802.1p/Q

IP Telephony Protocol

SIP (RFC 3261) over UDP, TCP, and TLS IMS (3GPP TS 24.229) RTP (RFC 3550) SDP (RFC 4566) Multi-part body support Redundancy support via DNS SRV Multiple trunk support Survivability for IP-Phones IPv4 and IPv6 dual stack signaling and media

Digital Telephony

Euro ISDN EDSS-1/ETSI PRI/NET5 BRI/NET3 ISDN NI-2 (US T1 PRI) ISDN DMS100 (US T1 PRI) ISDN 5ESS (US T1 PRI) ISDN speech, audio, and data (Fax Gr 4, UDI 64, and RDI 64) ECMA-143 (QSIG-BC) E1 R2 digital line signaling (ITU-T Q.421) E1 R2 MFC interregister signaling (ITU-T Q.441) Presets for: Brazil, Argentina, Mexico, Saudi Arabia, Venezuela, Philipines, and ITU-T T1/E1 E&M (Immediate, Wink-Start, Feature Group-B, and Feature Group-D), MF-R1, DTMF Advice of Charge AOC-D and AOC-E (ETS 300 182)

Analog Telephony

Support for call forward, call transfer, conference call, call waiting, CCNR, and CCBS Multiple country presets Customisable tones and ring patterns Echo cancellation Message Waiting Indication (MWI), via FSK and voltage

Caller ID detection (name & number) as per Bell-core

On-hook/off-hook caller ID generation (name & number) as per Bell-core DTMF or FSK and Telebras **BINA**

Answer and disconnect signaling

Call Routing

Local switching Call filtering and blocking Calling/called number manipulation using regular expressions Routing Criteria:

- Interface
- Calling/called party number
- Calling/called URI
- Time of day, day of week, and date
- Many others

Mapping and transformation of call properties to/ from SIP headers Hunt groups

Fax and Modem Support

Group 3/super G3 fax real-time fax over IP T.38 fax relay (9.6 k and 14.4 k) Clear channel (G.711) fax and modem pass-through

Networking

IPv4 – IPv6
Multiple IP addresses per link or VLAN
Multiple VLANs per link
DHCP client
PPPoE (RFC 2516)
IEEE 802.1q + DSCP QoS tagging (media, signaling, and mgmt)
IEEE 802.1x wired authentication
LLDP-med (ANSI/TIA-1057)
QoS traffic shaping
Firewall with stateful inspection, rate-limitation, and automatic black-listing
Static routing

Physical Interfaces

NAPT

DHCP Server

5 x 10/100/1000 Base-T Ethernet RJ-45 connectors 2 x TDM sync RJ-45 connectors 1 to 4 x RJ-48 E1/T1 connectors* 4, 8, 12, 16, 20, or 24 x RJ-48 BRI S/T connectors* 4, 8, 12, 16, 20, or 24 x RJ-11 FXS connectors*

4, 8, 12, 16, 20, or 24 x RJ-11 FXO connectors* (*Depending on configuration)

Power Supply

Internal 100-240 VAC power supply

Dimensions

Height: 4.4 cm

Width (mounting brackets): 48.5 cm

Depth: 19.5 cm Weight: 3Kg

Operating Environment

Operating temperature: 0°C to 40°C Storage temperature: -20°C to 70°C Humidity: up to 85%, non-condensing

Digital Ports	Up to 4 E1/T1 Up to 24 BRI
Analog Ports	Up to 24 FXS Up to 24 FXO
Mounting	Rack
Network	5 x 10/100/1000 Base-T
Survivability	✓



This datasheet applies to model: M.



Media5 Corporation is a global supplier of multimedia communication solutions, offering a complete set of IP-based products and technologies.

With a focus on innovation and excellence in customer support, we deliver highly adaptive hardware and ready-to-market software components. This allows our customers and partners to take advantage of secure, reliable, and comprehensive communication solutions.

Present in more than one hundred countries, Media5 has its headquarters in Canada and local representatives in North and Latin America, Europe, and the Middle East.

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