

# Mediatrix<sup>®</sup> Sentinel 100 SBC

The Sentinel 100 Hybrid SBC combines a Session Border Controller and a Media Gateway in a robust multi-service business platform capable of addressing complex needs for SMBs. Sentinel 100 goes up to 120 simultaneous VoIP channels and targets applications for up to 500 users.

With a flexible architecture designed for a variety of use cases, including SIP Trunking, Hosted Services, and Unified Communications, Sentinel 100 is a trustworthy solution for any stage of SIP deployment with QoS monitoring, security, survivability, and interoperability.

Featuring flexible support for digital E1/T1 PRI, ISDN BRI S0/T0, CAS (R2, E&M) and analog FXS/FXO interfaces, Sentinel 100 is also the most reliable platform for legacy PBX connectivity and fallback to the PSTN.



#### **Survivability**

Sentinel ensures service continuity by establishing external calls through the PSTN and by routing internal calls when the primary network is temporarily unavailable.

#### **Remote Users**

Sentinel solves far-end NAT traversal problems and support SIP manipulations, providing communication service access to branch offices, home workers, or travellers as if they were on the same site.

#### **Network Demarcation**

Create a clear separation between the enterprise's and the operator's networks by hiding the topologies and credentials, and by blocking unauthorised users.

#### **Legacy and IP Systems Integration**

With its flexible configuration of E1/T1 PRI, ISDN BRI, and FXS/FXO telephony ports, call-switching, and user-defined call properties (including caller/calling ID), the Mediatrix Sentinel smoothly integrates legacy CPE into IP systems.

# Mediatrix® Sentinel 100 SBC

# **Applications**

#### **Operators**

- ✓ Facilitate TDM replacement project by integrating the current legacy equipment in a first phase, and then complete the transition to pure SIP in a later phase, without having to replace the CPE.
- Ensure security, protection against fraud, and interoperability for SIP trunk deployments.
- Provide survivability in Hosted Unified Communications/PBX deployments by using a built-in PSTN gateway in case of a WAN failure.

#### **System Integrators**

- ✓ Safely connect SIP trunks, PSTN, branch offices, and mobile workers to the enterprise TDM or IP-PBX.
- ✓ Allow cost-effective and profitable SIP deployments in a smooth integration with SIP-based and legacy telephone systems.
- Create a shield of confidentiality between the enterprise and the Internet.

# **Key Features**

#### **Carrier-Grade Features**

T.38 and clear channel fax over IP

High performance processing of up to 120 voice channels

#### **Robust Security**

Enterprise communication encryption
SIP-enabled firewall inspects and authorises
communications and prevents DoS attacks

#### **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 rule-based SBC with dynamic routing and manipulations for solving complex deployment scenarios

# **Technical Specifications**

#### **Session Border Controller**

Back-to-Back user agent

SIP header manipulation

SIP registrar

SIP authentication

SIP failover

Registration throttling/caching

Call forking

Advanced, rule-based, call routing

Dynamic call routing based on:

- Peer monitoring state
- Registration cache

Call Admission Control (CAC), per trunk, based on:

- Call volume
- · Bandwidth usage
- Concurrent calls

Near and far-end NAT traversal

Audio and video media relay

Codec filtering

SIP and media encryption

UDP/TCP/TLS interworking

DTMF interworking

#### **Media Processing**

G.711 (A-law and  $\mu$ -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**

Signaling and media topology hiding

Denial of Service (DoS) protection of core and enter-

prise networks

Call rate limitation

SIP over TLS

SRTP with AES cipher - 128 bits

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-

384

#### 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)

Subscriber's active registration and call monitoring

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

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 inter-register 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

(80v)

Caller ID detection (name & number) as per Bell-core FSK On-hook/off-hook caller ID generation (name & number) as per Bell-core DTMF or FSK and Telebras BINA Answer and disconnect signaling

#### **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

NAPT

**DHCP Server** 

#### **Operating Environment**

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

#### **Power Supply**

Internal 100-240 VAC power supply

#### **Physical Interfaces**

5 x 10/100/1000 BaseT Ethernet RJ-45 connectors

2 x TDM sync RJ-45 connectors

1 x USB 2.0 Type-A connector

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)

#### **Dimensions**

Height: 4.4 cm

Width (mounting brackets): 48.3 cm

Depth: 19.5 cm

#### **SBC Licensing**

A license is needed for each concurrent call.

SBC Sessions	Up to 120
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

This datasheet applies to model: M.



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