

Hybrid E-SBC and Media Gateway

# **UNIVERGE BX800**



# **Benefits**

- · Fully integrated device for secured SIP trunking and
- Hybrid SBC and Media Gateway platform lowers CAPEX and reduces space and power footprints
- · Extensive interoperability and partnerships that extend across multiple vendor devices and protocol implementations
- · Offers comprehensive security, interoperability and reliability
- · Delivers high service performance and voice quality
- · Branch office survivability in the event of a WAN outag

# Key features

- · Rich and powerful SIP normalization and routing mechanisms for seamless interoperability
- · Hybrid SBC that connects to PSTN / PBX trunks for fallback and gradual enterprise migration to SIP
- Support for analog (FXO, FXS) and digital (PRI, BRI)
- · Perimeter defense against denial of service, fraud and eavesdropping
- · VoIP quality monitoring and enforcement
- · High Availability using two box redundancy
- Media Processing for Transcoding, Gain Control, DTMF/Fax, etc.

The NEC UNIVERGE BX800 Enterprise Session Border Controller (E-SBC) and Media Gateway offers a complete connectivity solution for small-to-medium sized enterprises.

The BX800 connects IP-PBXs to any SIP trunking service provider, scaling up to 250 concurrent SBC sessions. It offers superior performance in connecting any SIP to SIP environment, legacy TDM-based PBX systems to IP networks, and IP-PBXs to the PSTN, supporting up to 60 voice channels in a 1U platform.

# Vast mediation capabilities and proven interoperability

The BX800 supports a wide range of voice coders and is capable of transcoding between narrowband and wideband voice coders, providing SIP normalization, fax handling, gain control and numerous additional media processing features. It offers certified interoperability with leading unified communications solutions and SIP trunking providers.

#### Security

The BX800 provides robust protection for the IP communications infrastructure, preventing Denial of Service, fraud and service theft and guarding against cyber-attacks and other service-impacting events.

# Reliability

The BX800 offers active/standby high availability and maintains high voice quality to deliver reliable enterprise VoIP communications. Advanced call routing mechanisms, network voice quality monitoring and branch survivability capabilities (including PSTN fallback with E911) result in minimum communications downtime.

### **Applications**

- SIP trunking
- · Hosted PBX & UC as a Service
- · Remote and mobile worker support
- SIP mediation between UC and IP-PBX systems









# **Specifications**

Capacities  Max. Signaling/Media Sessions	250	Max. SRTP/RTP Sessions	250
Max. Transcoding Sessions	57	Max. Registered Users	1500
Telephony Interfaces		3	
nalog	4 FXS/FXO interfaces		
Digital	Two E1/T1 interfaces with an option for PSTN Fallback / 4 Basic Rate interfaces		
Digital PSTN Protocols	Supporting various ISDN PRI protocols such as EuroISDN, North American NI-2, Lucent™ 4/5ESS, Nortel™ DMS-100 and others. It also		
3	supports different variants of CAS protoco	ols, including MFC R2, E&M immediate st	art, E&M delay dial / start and others
Network Interfaces			
Ethernet	4GE or 4GE + 8FE interfaces configured in	1+1 redundancy or as individual port	
Security			
Access Control	DoS/DDoS line rate protection, bandwidth throttling, dynamic blacklisting		
olP Firewall	RTP pinhole management, rogue RTP detection and prevention, SIP message policy, advanced RTP latching		
ncryption/Authentication	TLS, DTLS, SRTP, HTTPS, SSH, client/server SIP Digest authentication, RADIUS Digest		
Privacy	Topology hiding, user privacy		
raffic Separation	VLAN/physical interface separation for mu	ultiple media, control and OAMP interface	es
ntrusion Detection	System Detection and prevention of VoIP attacks, theft of service and unauthorized access		
Interoperability			
SIP B2BUA	Full SIP transparency, mature and broadly deployed SIP stack, stateful proxy mode		
SIP interworking	3xx redirect, REFER, PRACK, session timer, early media, call hold, delayed offer		
Registration and Authentication	User registration restriction control, registration and authentication on behalf of users, SIP authentication server for SBC users		
ransport Mediation	SIP over UDP/TCP/TLS/WebSocket, IPv4 / IPv6, RTP / SRTP (SDES/DTLS)		
Message Manipulation	Ability to add/modify/delete SIP headers and message body using advanced regular expressions (regex)		
JRI and Number Manipulations	URI user and host name manipulations, ingress and egress digit manipulation		
ranscoding and Vocoders	Coder normalization including transcoding, coder enforcement and re-prioritization, extensive vocoder support: G.711, G.723.1, G.726		
	G.729, GSM-FR, AMR-NB/WB, AMR-NB, AMR-WB (G.722.2), SILK-NB/WB, Opus-NB/WB		
Signal Conversion	DTMF/RFC 2833/SIP, T.38 fax, T.38 V3, V.34, packet-time conversion V150.1		
VebRTC Controller	Interworking between WebRTC devices and SIP networks Supports WebSocket, Opus, VP8 video coder, lite ICE, DTLS, RTP multiplexin		
IAT.	secure RTCP with feedback		
NAT	Local and far-end NAT traversal for suppo	rt of remote workers	
Voice Quality and SLA	December of Silling of Secretariation	al ada a sala a fara a fara a fara da	
Call Admission Control	Based on bandwidth, session establishme	ent rate, number of connections/registrati	ons
Packet marking	802.1p/Q VLAN tagging, DiffServ, TOS		
Standalone Survivability	Maintains local calls in the event of WAN failure. Outbound calls can use PSTN fallback for external connectivity (including E911)		
mpairment Mitigation	Packet Loss Concealment, Dynamic Programmable Jitter Buffer, Silence Suppression/Comfort, Noise Generation, RTP redundancy, broken connection detection		
/oice Enhancement	Transrating, RTCP-XR, Acoustic echo cancellation, replacing voice profile due to impairment detection, Fixed & dynamic voice gan contr		
Direct Media	Hair-pinning of local calls to avoid unnece	ssary media delays and bandwidth consi	umption
No Media Anchoring)	DTOD VD		
/oice Quality Monitoring	RTCP-XR  SRC high availability with two-box redundancy, active calls presented		
High Availability (Redundancy)	SBC high availability with two-box redundancy, active calls preserved  Access control and media quality enhancements based on QoE and bandwidth utilization		
Quality of Experience	Ability to remotely verify connectivity, voice quality and SIP message flow between SIP UAs		
est agent	Ability to remotely verily connectivity, voic	e quality and SIP message now between	SIF UAS
SIP Routing			
Routing Methods	Request URL, IP address, FQDN, ENUM, advanced LDAP, third-party routing control through REST API		
dvanced Routing Criteria	QoE, bandwidth, SIP message (SIP request, coder type, etc.), Layer-3 parameters		
Routing Features	Least-cost routing, call forking, load balancing, E911 gateway support, emergency call detection and prioritization		
SIPRec	IETF standard SIP recording interface		
Management			
DAM&P	Browser-based GUI, CLI, SNMP, INI Conf	iguration file, REST API	
Physical / Environmental			
Dimensions	1U x 320mm x 345mm (HxWxD)	Weight	Approx. 5.95lb (2.7kg) loaded with OSN
Mounting	Desktop or 19" rack mount	Power	100-240V 4A 50-60 Hz

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