

Session Border Controller

UNIVERGE BX9000



Benefits

- Meets demands for datacenter infrastructure harmonization and NFV
- Supports Network Functions Virtualization (NFV)
- Offers comprehensive security, interoperability and reliability
- Flexible licensing options for cost-effective scalability
- Delivers high service performance and voice quality

Key features

- · Hypervisors: VMware, Hyper-V and KVM
- Scalable to thousands of SBC sessions
- Extensive SIP mediation capabilities
- Supports remote workers and mobile SIP clients
- · Perimeter defense against denial of service, fraud and eavesdropping
- · VoIP quality monitoring and enforcement
- · Branch survivability during WAN failure
- · Active/Standby High Availability

The UNIVERGE BX9000 is a sofware Enterprise Session Border Controller (E-SBC), designed to offer enterprises a flexible and scalable SBC solution that meets the requirements of today's data center infrastructures. The UNIVERGE BX9000 supports flexible SIP interoperability, delivering service assurance and enabling scalable, reliable and secured connectivity between different VoIP networks.

Extensive Mediation Capabilities and Proven Interoperability

The UNIVERGE BX9000 includes comprehensive media security and SIP normalization capabilities. It offers full interoperability with an extensive list of IP-PBXs, unified communications solutions and SIP trunking provider networks.

Security

The UNIVERGE BX9000 provides robust protection for the IP communications infrastructure, preventing fraud and service theft and guarding against cyber-attacks and other service-impacting events.

Reliability

The UNIVERGE BX9000 offers active/standby high availability and maintains voice quality to deliver reliable enterprise VoIP communications. Advanced call routing mechanisms, network voice quality monitoring and branch survivability capabilities result in minimum communications downtime.

Applications

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









Specifications

| Capacities | |
|----------------------------------|---|
| Max. Signaling Sessions | 24,000 |
| Max. SRTP-RTP Sessions | 10,000 |
| Max. Transcoding | 1,200 |
| Max. Registered Users | 75,000 |
| Security | |
| Access Control | DoS/DDoS line rate protection, bandwidth throttling, dynamic blacklisting (Intrusion Detection System) |
| VoIP Firewall | RTP pinhole management, rogue RTP detection and prevention, SIP message policy, advanced RTP latching |
| Encryption/Authentication | TLS, DTLS, SRTP, HTTPS, SSH, client/server SIP Digest authentication, RADIUS Digest |
| Privacy | Topology hiding, user privacy |
| Traffic Separation | VLAN/physical interface separation for multiple media, control and OAMP interfaces |
| 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 and more |
| Registration and Authentication | User registration restriction control, registration and authentication on behalf of users, SIP authentication server for SBC users |
| Transport Mediation | SIP over UDP/TCP/TLS/WebSocket, IPv4 / IPv6, RTP / SRTP (SDES/DTLS) |
| Message Manipulation | Add/modify/delete SIP headers and message body using simple WireShark-like language with powerful capabilities such as variables |
| | and utility functions |
| URI and Number Manipulations | URI user and host name manipulations, ingress and egress digit manipulation |
| Transcoding 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, AMR-WB (G.722.2), SILK-NB/WB, Opus-NB/WB |
| Signal Conversion | DTMF/RFC 2833/SIP, T.38 fax, packet-time conversion |
| WebRTC Controller | Interworking between WebRTC devices and SIP networks, Supports WebSocket, Opus, VP8 video coder, lite ICE, DTLS, RTP multiplexing, secure RTCP with feedback |
| Voice Quality and SLA | Local and far-end NAT traversal for support of remote workers |
| Call Admission Control | Based on bandwidth, session establishment rate, number of connections/registrations |
| Packet marking | 802.1p/Q VLAN tagging, DiffServ, TOS |
| Standalone Survivability | Maintains local calls in the event of WAN failure |
| Impairment Mitigation | Packet Loss Concealment, Dynamic Programmable Jitter Buffer, Silence Suppression/Comfort Noise Generation, RTP redundancy, broken connection detection |
| Voice Monitoring and Enhancement | Transrating, RTCP-XR, Acoustic echo cancellation, replacing voice profile due to impairment detection, Fixed & dynamic voice gain control, packet loss concealment, dynamic programmable jitter buffer, silence suppression/comfort noise generation, RTP redundancy, broken connection detection |
| Direct Media | Hair-pinning (No Media Anchoring)of local calls to avoid unnecessary media delays and bandwidth consumption |
| High Availability (Redundancy) | SBC high availability with two-box redundancy, active calls preserved |
| Quality of Experience | Access control and media quality enhancements based on QoE and bandwidth utilization |
| Test agent | Ability to remotely verify connectivity, voice quality and SIP message flow between SIP UAs |
| SIP Routing | |
| Routing Methods | Request URL, IP address, FQDN, ENUM, advanced LDAP, third-party routing control through REST API |
| Advanced 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 | |
| OAM&P | Browser-based GUI, CLI, SNMP, INI Configuration file, REST API OVOC, HTTP reverse proxy |
| Multi Tenancy | Advanced multi-tenant SBC partitioning |
| Minimum Requirements | |
| Hypervisor | VMware® vSphere ESXi™, Linux KVM, Microsoft Hyper-V |
| Memory | 2 GB |
| Disk space | 10 GB |
| Virtual NICs | 2 (Standalone) / 3 (High Availability) |
| Virtual CPUs | 2 |
| | |

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North America (USA, Canada) NEC Corporation of America www.necam.com

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