

High Availability

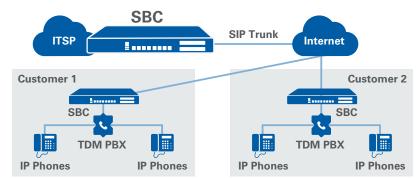
Ensure business continuity with our High availability (HA) feature, allowing mirroring of your main SBC with a standby SBC ready to automatically take over calls in case of failure. This feature is included free of charge!

Protection from Enterprise Security Threats

- Denial of Services Call / registration overload | Malformed messages (fuzzing)
- Configuration Errors Mis-configured devices | Operator and application errors
- •Theft of Service / Fraud Unauthorized users | Unauthorized media types
- BYOD Smartphones running unauthorized apps | Viruses and malware attacking your VoIP network

Use Cases

- Service Provider Applications -
 - SIP trunking
 - Remote worker
 - -- Hosted PBX
 - -- Core session router and load balancer





✓ Field Upgradeable Session Expansion

✓ Browser-based GUI for Easy Configuration

✓ Cost-effective Session-based Licensing

Session-based Licensing – No Hidden Fees

✓ PSTN Equivalency for SIP Trunking Deployments

✓ RAID-1 Solid State Drives

✓ Hardware-based Transcoding & Media Handling

✓ Optional Annual Support & Software Maintenance Plans



Technical Specifications

Capacities

- Max. 4,000 calls / sessions
- · Full transcoding capability
- Full security (SRTP / TLS) capability
- · Unlimited SIP trunks

Media Capabilities

- · Voice, Video, FAX, IM and Presence support
- Full RTP transcoding (G.711, G.722, G.729, G.726, G.723.1, iLBC, AMR, G.722.1)
 *Only with hardware transcoding option
- T.38 Fax Relay *Only with hardware transcoding option

Networking

- 6x 10/100/1000 BaseT Ethernet ports
- IPV4, IPV6
- VLAN support
- 1x console port
- 1x VGA port

Licensing

- · Only max. number of calls are licensed
- Field upgradeable in 250/500/1000/2000 call increments

Security

- DDoS / DoS attack protection
- · Call security with TLS / SRTP
- · Media security with SRTP
- IPSec encryption
- · IP firewall with port forwarding
- Two-stage authentication

High Availability / Redundancy

 1:1 active / standby two-box redundancy for business continuity

Call Control

- · Advanced XML routing engine
- · Least cost routing
- · Rate limiting: Call and registration
- Endpoint authentication
- · Media bandwidth policy
- · Intelligent media anchoring / release

VoIP

- SIP 2.0 compliant
- · SIP trunking and remote working
- SIP intrusion prevention
- · SIP registration scan attack detection
- SIP request rate limiting
- · SIP registration pass-thru
- SIP header normalization
- SIP malformed packet protection
- Advanced NAT traversal capabilities
- Topology hiding
- ENUM routing

Session and Monitoring

- Multiple session routing options
- RTCP statistics reports
- QoS (ToS or DSCP)
- RADIUS CDR and authentication
- · QoS monitoring and reporting

Debugging

- Dedicated Browser interface for capturing full RTP media and signaling information
- Onboard browser-based PCAP tracing, signaling and media – wireshark compatible
- Large onboard storage capacity for long term tracing

Management

- · Easy to use web interface
- HTTP XML-based CDR
- · Real-time monitoring and debugging
- TR-069 for remote provisioning
- REST based interface to remotely configure SBCs

Telephony Interface

- Up to 4 E1/T1s (with A101DE/A102DE/A104DE card)
- Echo cancellation up to 128ms tail
- Primary rate ISDN, configurable NT/TE
- T1: Q.931, NI2, 4ESS, 5ESS, DMS-100
- E1: Q.931, ETSI

Hardware Specifications

- Power: Dual internal 100~240V power supply
- Dimensions: 430mm (W) x 520mm (D) x 44.5mm (H)
- Weight: 6.35kgs (14lbs)
- 19 inch rackmount brackets included

