

MyPBX U300

Embedded Hybrid IP-PBX for Small Business



IVR



Conference



RingGroup



Queue



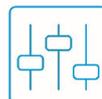
Voicemail



PIN User



Firewall



QoS



VLAN



Redundancy

MyPBX U300 boasts an embedded PRI (E1/T1/J1) port and 2 FXS ports in the one compact system, providing higher density trunking for offices using E1 PRI signaling. This system supports up to 300 users and 50 concurrent calls. Also it could be used as a gateway to legacy PBX systems in applications.

- **Easy to use**
Easy to deploy and manage via web-based configuration interface
- **Robust all-in-one features**
Deliver enterprise-class communication features and functionality to SMBs
- **Redundancy**
Support hot standby settings (automatic failover) in case of server failure
- **Match your IP phone**
Perfect interoperability with a wide range IP Phone
- **Speak your language**
Multi-language Web interface and voice prompts
- **No future licensing fees**
Scalable with plug-and-play ease without licensing
- **Used as a gateway**
Could be used as a E1 gateway (E1 to VoIP or VoIP to E1)
- **Energy Saving**
Embedded system with low power consumption for your green office



PBX Features

- Automated Attendant (IVR)
- Attended Transfer
- Blind Transfer
- Blacklist
- Call Back
- Call Detail Records (CDR)
- Call Forward
- Call Parking
- Call Pickup
- Call Routing
- Call Waiting
- Caller ID
- Conference
- Do Not Disturb (DND)
- Follow me
- Intercom/Zone Intercom
- Music on Hold
- Music on Transfer
- Queue
- Ring Group
- Skype Integration (Skype Connect)
- Speed Dial
- Voicemail (3000 minutes)
- Voicemail to email
- Voicemail Forwarding

Advanced Features

- Dial by Name
- Direct Inward System Access (DISA)
- Distinctive Ringtone
- LDAP Server
- Mobility Extension
- One Touch Recording
- Phone Provisioning for Aastra, Cisco, Escene, Fanvil, Grandstream, Htek, Panasonic, Polycom, Snom, Yealink IP Phones
- QoS (voice quality)
- Route by Caller ID
- Spy Functions
- Static Route

Add-on

- Billing System
- Call Recording
- Hotel Module
- MyPBX Client

Hardware Interface

- 1 LAN Ethernet 10/100 BASE-T
- 1 WAN Ethernet 10/100 BASE-T
- 1 E1/T1/J1 port (support PRI, MFC R2, SS7, E&M)
- 2 FXS ports
- 1 USB port (2.0)
- 1 RS232 port
- 1 Audio Input port
- 1 Audio Output port

System Capacity

- 300 IP phone users
- 50 concurrent calls
- 512 MB Onboard Flash
- 512 MB Onboard RAM

Voice Processing

- Protocol: SIP (RFC3261), IAX2
- Transport: UDP, TCP, TLS, SRTP
- DTMF: RFC2833, SIP INFO, In-band
- Codec: G.711 (a-law, u-law), G.722, G.726, G.729 A, GSM, Speex, ADPCM, H261, H263, H263p, H264, MPEG4

Multiple Languages

- System voice prompt: American English, Australian English, British English, Afrikaans, Arabic, Chinese, Dutch, French, Canadian French, Danish, Finnish, German, Greek, Hebrew, Hungarian, Italian, Korean, Norwegian, Persian, Polish, Portuguese, Brazilian Portuguese, Russian, Spanish, Latin American Spanish, Mexican Spanish, Thai, Turkish, Swedish

- Web GUI: English, Chinese Simplified, Chinese traditional, Portuguese, Spanish, Russian, Hebrew, Turkish, French, Italian, Polish, Romanian, Albanian, Thai, Korean, Persian, German, Dutch, Croatia

Security

- Firewall
- SIP TLS transport
- SRTP (RTP encryption)

Faxes

- T30, T38 faxes
- Fax to Email
- Incoming fax tone detection

Internet

- DHCP server
- DDNS
- Static IP
- DHCP client
- PPPoE
- VLAN
- VPN: OpenVPN, L2TP, IP Sec

Physical

- Size: 213x160x44 mm
- Weight: 1.2 kg
- Power Supply: AC 100-240V (DC 12V 2A)
- Operation Range: 0 to 40°C, 32 to 104°F
- Storage Range: -20 to 65°C, -4 to 149°F
- Humidity: 10-90% non-condensing