#Luke_{.#}



50 SIP extensions > 2 FXO trunks

12 Concurrent Calls > 2 FXS extensions

Auto-attendant
> 3 SIP trunks



Supported standards:

SIP RFC 3261 Compliance

Voice Codecs: G.711 a/ μ , G.729A and G.723.1,

G.726-32K, GSM

T.38, T.30 FAX support

Video Codec Pass-Thru: H.263, H.264

MIME, SMIME authentication

RTP, sRTP

Internet: DHCP, Static IP, PPPoE, PPtP

Network: NAT & DHCP server

Call Features

Call Hold, Call Waiting, Call Parking, Call Pickup

Blind Call Transfer, Attended Call Transfer

Call Forward on Busy, No Answer or Always

Do Not Disturb

Speed Dial

Outgoing & Incoming call blocking

Hot Line, Warm Line

Ring Groups & Hunt Groups

Direct Inward Dialing (DID)

Call Queuing

3-way calling

Call Logging, Call Reporting

Busy Lamp Field (BLF) per RFC4235

Voice Mail

Personal greeting

Multi-Languages support

MWI notification (option)

Email notification

Redirect to PBX /SIP/ PSTN

Auto Attendant

Web-Based Auto Attendant Flow Editor

Scheduled Special Announcement

Holidays Working Time Support

Multiple Language Support

Call Conference

Conference Calling

Meeting Rooms

Up-to 6 Participants per Room

Management

Web-Based management

User and administration configuration mode

NTP synchronization

System event Syslog

Scheduled Backup

Firewall and DMZ

Access control

USB device Restore and Backup configuration

VXML Based IVR

Physical

Two WAN ports, for connecting to Internet

One LAN port for connecting to Private Network

One USB port

4 RJ-11 ports, 2 for FXS and 2 for FXO

IEEE802.11n, 2T2R

Dimension: W 22.2x H 3.3x D 14.5 cm (excl. Stand)

Weight: 450g (without packing)

One unit packing dimension: 55 x 37.5 17.5 CM

Environmental

0°C~45°C,

10%~90% relative humidity, non-condensing