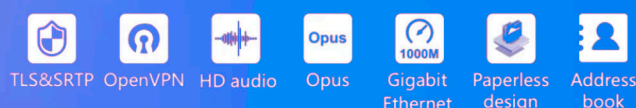




SIP-R80

- ✓ 2.8-inch colored LCD display screen
- ✓ 8 SIP lines, paperless electronic page turning
- ✓ Encrypted call (OpenVPN/TLS/SRTP)
- ✓ Noise reduction HD sound quality
- ✓ Can be equipped with USB recording/Wi-Fi
- ✓ Advanced remote address book management
- ✓ Elite business style, aesthetic and practical design



Explore the SIP-R80 Series - Communication Excellence

The SIP-R80 is an innovative, high-end business phone designed specifically for management elites and professionals. Its business like appearance combined with HD color screen brings an unprecedented visual display. This phone redefines the user experience with its smooth operation interface, fast response speed and extensive functions. It aims to greatly enhance teamwork and personal effectiveness through excellent communication skills.

Brightening characteristic

The art of sound: With its ultra HD sound quality and Opus broadband audio codec support, it delivers an unparalleled audio calling experience that brings every conversation to life.

Power of connection: Dual 100 megabit network ports with PoE power supply as well as the integration of USB interface ensure a high degree of connectivity and convenience of the device.

A new level of vision: Equipped with a 2.8-inch backlit LCD display with a resolution of up to 320x240, it brings a clear and bright visual experience for every view and operation.

Efficient office, at your fingertips: Paperless programmable button design helps farewell to cumbersome menu operations. Touching the button to quickly access common functions makes efficient office easy to achieve without complex operations.

Rich features and excellent performance: Supporting 8 SIP accounts and multiline calls, compatible with five-party conference and a range of SIP call functions, the SIP-R80 is perfectly suited to various enterprise communication needs.

Phone function

- 8 SIP lines
- Multi-language selection (Chinese/English/Turkish/Korean/Russian/Traditional Chinese/French/Italian/German/Portuguese/Polish/Thai/Hindi/Arabic/Japanese/Spanish/Vietnamese)
- Caller ID with display name
- Custom DSS key, BLF monitoring, hotline, Speed dial
- Handle/hands-free/headphone mode
- Call hold, mute, blind transfer, inquiry transfer
- Do not disturb, call forwarding, call waiting
- Multicast, intercom, text message, keypad lock, emergency call
- IP direct call, redial, callback, auto answer
- Five-party conference, line sharing, dialing rules
- 10 types of built-in ringtones / upload custom ringtones
- Manually or automatically synchronize network time
- Action URL & Action URI

Audio characteristic

- High-fidelity sound quality: HD handset, HD handset-free
- Broadband codec: G.722, Opus
- Narrowband codec: G.711(A/μ), G.723.1, G.729AB, G.726, iLBC
- DTMF: In-band transmission (In-band), Out-of-band transmission (RFC 2833), SIP INFO
- Full-duplex hands-free
- Acoustic Echo Cancellation (AEC)
- Voice Activity Detection (VAD)
- Comfort noise generation (CNG)
- Packet loss compensation (PLC)
- Background noise detection (BNE)
- Automatic gain control (AGC)
- Dynamic adaptive RTP jitter buffer up to 300ms

Address book

- Blacklist
- XML/LDAP address book
- Contact smart search
- Contact export/import
- Local contacts (1000 entries can be stored)
- Call history: dialed/received/missed/forwarded calls (1200 records in total)

IP PBX Features

- BLF, voice message
- Anonymous call, anonymous call rejection
- Call park, call pickup
- Server-based recording function

Configuration and maintenance

- Web configuration, Phone configuration
- Automatic deployment: Support FTP/TFTP/HTTP/HTTPS/SIP PnP/TR-069
- BIN configuration
- Contact configuration export/import
- CSTA remote control
- Network packet capture and diagnosis
- Custom/full restore factory settings
- System firmware upgrade

Network characteristics

- Support IPv4 / IPv6 / IPv4&IPv6
- IP address allocation mode: static IP / DHCP
- Virtual Local Area Network: VLAN
- Data link layer protocol: CDP / LLDP
- SIP v1 (RFC2543), v2 (RFC3261)
- Support redundant servers
- Support ICE, STUN (NAT session traversal)
- Support UDP / TCP / TLS transmission
- Support SRTP encryption protocol
- SIP trust server (anti-attack mechanism)
- HTTP/HTTPS
- TR-069 network management protocol
- SNTP automatically synchronizes network dates and times
- QoS: QoS: 802.1p/Q tagging (VLAN), Layer3/ToS/DSCP
- HTTPS certificate management
- Support configuration file encryption
- Support information verification mechanism
- RTCP-XR, VQ-RTCPXR
- OpenVPN, IEEE802.1X

Phone interface

- 1 x RJ-9 (4P4C) handset interface
- 1 x RJ-9 (4P4C) headset interface
- 2 x RJ-45 10/100M adaptive Ethernet port
- DC power interface
- USB2.0 interface

Product specification

- Phone main screen: 2.8-inch color LCD display
- Information indicator: LED red light
- Phone keyboard:
 - 8 SIP line keys (with dual-color LED lights, programmable)
 - 4 customizable soft keys
 - 12 standard telephone number keys
 - 6 function keys: headset, voice message, mute, Call forwarding, redialing, hands-free
 - 5 navigation keys
 - 2 volume up/down adjustment keys
- Color: business gray
- Installation method
 - Desktop
 - Wall-mounted
- Desktop size: 187mm*184mm*147mm
- Phone power consumption: 2~3W
- PoE: IEEE 802.3af/IEEE 802.3at
- Working environment humidity: 10~95%
- Working temperature:
 - 10~50° C (+14~122° F)

Contact Us

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