

Entry Level SIP Phone

AT-810



RELIABLE

- Customized ring tone for different regions
- Independent RJ9 headset support
- Wall mounted structure
- Full duplex speaker with HIFI voice quality
- Support HD Voice
- Back light on LCD display

VOIP

- Sip 2.0 (RFC3261)
- 1SIP Accounts
- Support SIP domain,SIP authentication(none,basic,MD5)
- Support DTMF:Inband Audio,RFC2833,SIP Info
- Support Pickup,Call Park,Auto answer,Paging,Intercom Call Forward,Transfer,Call hold,Call waiting, 3-way talking
- Support DND,Redail,Mute,
- Dial-peer calling rule, IP to IP call
- Phone book with 100 records,
- 50 dialed/answered/missed calls
- Support speed dial,Voice mail

NETWORKING

- LAN/PC: Support bridge mode
- LAN support DHCP,Static and PPPoE
- Support VLAN (DATA VLAN and VOICE VLAN)
- Support L2TP VPN without IPSec
- Qos support diffserv for SIP signaling and voice stream(RTP&RTCP)
- Support ToS
- Support SNTP client
- Support SRTP(RFC3711)
- Support TLS(Transport Layer Security)

VOICE

- Support G.711A/u, G722, G.723, G.726-16/24/32/40 iLBC,G.729,
- G.168 compliant 96ms echo cancellation
- VAD,AEC,AGC,CNG,AJB
- Support HAC (Hearing Aids Compatibility)

MANAGEMENT

- Support different level user management
- Configuration via web , keyboard
- Firmware and configuration updating via HTTP , FTP and TFTP
- Support syslog
- Support auto provision for configuration file upgrade

HARDWARE

- LCD: Segment LCD
- FLASH: 8M
- RAM: 8×16M
- Dual Core CPU
- Support PoE(AT810P Only)

Certifications

- CE: EN55024 EN55022
- RoHS
- Elastix