



Hotel Compact Type IP Phone

S1811



S1811 hotel IP phone can satisfy most hotel rooms and even the application requirements, beside, it has good expansibility and clear characteristics of quality voice calls.

S1811 is a popular type IP Phone in hotel phones series, with modern design, functional, practical and voice clarity characteristics. It accomplished the powerful telephony features by cooperating with the communications platform, such as call transfer, hotline, voice mail, interruption-free, etc.

S1811 IP phone supports defined 10 programmable keys ,and it can define according to the hotel different needs. such as hotel all kind of equipment service hotline (housekeeping, ticketing, switchboard, food and beverage, etc.), the hotel special features (alarm clock, voice mail, etc.)

Key Features

- Support 10 programmable keys.
- HD Voice: HD Codec, HD Speaker.
- WEB support Multi-language, e.g. Chinese, English, Russian, French etc.
- 2 SIP accounts.
- Support IEEE802.3af class 0(POE).
- Convenient and fast for Installation and maintenance
Support HTTP/TFTP/Telnet/Auto provision, and TR069 centralization maintenance.
- Call hold, Call waiting, Call forward, Call return, Call transfer.

S1811 Technical Datasheet

Phone Features

2 SIP accounts, Hotline
Call hold, Call waiting, Call forward, Call return, Redial, Mute, DND
Call transfer (blind/busy/ask)
Auto-answer, Speed dial, SMS, Voicemail
Message Waiting Indication LED
Tone scheme, Volume control
Direct IP call without SIP proxy
Ring tone selection/import/delete
Phonebook, Black list, Hand-free indicator.
Call history: Multi-Language
Support switching SIP server & Second server; 10 programmable keys

Advanced Features

Flash keys function:

>During a call, press the flash key to enter the transferred pattern with dial tone. The call will be holding.

>When transferring the dialing, pressing the flash key can cancel the transferred function. The call still be holding.

>Pressing the flash key can recover the on holding call

IP PBX System Integration

DND ; Forward synchronization
Intercom, Paging, Music on hold
Call pickup, Dial Plan,
Call recording Anonymous call,
Anonymous call rejection

Compatible with the platform

Codes and Voice Features

Wideband Codec: G.722
Narrowband codec: G.711μ/A, G.726, G.729AB, iLBC
VAD, CNG, AEC, AGC, Full-duplex,

Security

802.1x, VLAN QoS (802.1pq), SRTP
Transport Layer Security (TLS)
Digest authentication using MD5/MD5-session
Secure configuration file via AES encryption
Encryption for configuration file
Phone lock for personal privacy protection
Admin/User 2-level configuration mode , LLDP, CDP
L2TP VPN tunneling protocol - added

Network Features

SIP v1 (RFC2543), v2 (RFC3261)
SIP connection: Proxy mode & Peering Points mode;
SRTP voice encryption(RFC3711)
Redundant server support
NAT Traversal: STUN mode
DTMF: In-Band, RFC2833, SIP Info, Auto
HTTP/HTTPS Web Management
IP Assignment: Static/DHCP/PPPoE
Bridge/Router mode for PC port
TFTP/DHCP/PPPoE client
Telnet/HTTP/HTTPS server, DNS client,
NAT/DHCP server
Layer 3 ToS, DSCP

Management

Auto-provision via
FTP/TFTP/HTTP/HTTPS , TR069 optional
Configuration:
browser/phone/auto-provision
Trace package and system log export

Physical Features

30keys, including 12 digital keys
10x programmable keys, 1x redial key, 1x hold key, 1x speaker key, 1x flash key, 1x mute key
1x message LED
1x RJ-9 handset port
1xRJ45 10/100M Ethernet ports
Power over Ethernet ,IEEE 802.3af,class 0
Operating humidity: 10~95%
Storage temperature: up to 60°C
Power adapter: AC 100~240V input and DC 5V/1A output

Package Features

Qty/CTN:

N.W/CTN:

Carton Meas:

Certifications



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