

HS118V2 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.

HS118V2 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, multi-party conferences, voice mail, interruption-free, etc.

HS118V2 IP phones support defined 10 programmable keys ,and it can define according to the hotel different needs. such as hotel all kind of equipment sevice 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.
- One SIP accounts and support three-way conference.
- Support IEEE802.3af class 0(POE).
- Convenient and fast for Installation and maintenance Support HTTP/TFTP/Telnet/Auto provision.
- Call hold, Call waiting, Call forward, Call return, Call transfer.

HS118V2 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, 3-way conferencing

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

XML Phone Book search and input &

output

Enterprise phone book(800 entries)

LDAP phone book

Functions customizable keys

IP PBX System Integration

 ${\sf DND} \; ; \; {\sf Forward} \; {\sf synchronization} \;$

Intercom, Paging, Music on hold

Call pickup, Dial Plan,

Call recording Anonymous call,

Anonymous call rejection

Codes and Voice Features

Wideband Codec: G.722

Narrowband codec: G.711µ/A, G.723.1

G.726, G.729AB,iLBC

VAD, CNG, AEC, AGC, Full-duplex,

Security

VLAN QoS (802.1pq).

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

DNS SRV (RFC3263);

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

DNS client, NAT/DHCP server

Layer 3 ToS, DSCP

Management

Auto-provision via FTP/TFTP/HTTP/HTTPS

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

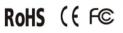
Package Features

Qty/CTN:

N.W/CTN:

Carton Meas:

Certifications





Compatible with the platform



























ESCENE COMMUNICATION CO.,LTD

Add: Floor 4 building C YueAn Industrial Park No.51 Huangcun Road Tianhe District, Guangzhou, China.

Tel: (86)20-82320720 Fax: (86)20-82326757