

H60P

HD Audio and Enterprise Entry Level
Hotel IP Phone



H60P Hotel Phone



Key Features

- HD Voice
- 2 SIP Accounts
- 6 programmable soft keys
- 5-way Conference
- Dual-port 10/100Mbps Ethernet
- Support for customized hotel specialty faceplate
- Full Compatible with 3CX, Asterisk, Yeastar, Xorcom, VoIP.ms, Broadsoft, FreeSWITCH Platform etc

Overview

The Dinstar H60P Hotel Phone is a cost-effective, multifunctional HD IP voice phone specifically designed for the hospitality industry, aimed at delivering an exceptional communication experience to clients. Its design is simple and easy to operate, compatible with various IP voice telephone systems, and can flexibly adapt to the diverse business needs of hotels.

The H60P phone integrates HD audio technology to ensure clear and smooth call quality. It supports 100 Mbps Ethernet and is compatible with PoE power supply, effectively reducing wiring costs. Additionally, it is equipped with six programmable keys that can be used for

personalized configuration of various hotel service hotlines. Moreover, the H60P supports customization of the phone faceplate according to customer needs, enhancing the user experience. By perfectly integrating with IPPBX, it offers a wide range of call service functions. The H60P supports features such as speed dial, call waiting, and voicemail, which are suitable for hotels, helping hotels enhance service quality and customer satisfaction.

Model	Screen	Network	PoE Support	Power Adapter
H60P	-	2*10/100M	√	Optional

Call Features

- 2 SIP Accounts
- Call Waiting
- Call Forward
- Blind/attended transfer
- Call hold, Call mute, DND mode
- Speed Dialing, Hotline
- Redial, Call Return
- Auto-redial, Auto-answer
- Dial Plan
- 5-way conference
- Peer Dialing

Phone Features

- Adjustable Voice Volume
- Selectable Ring Tones
- Action URL/URI
- Programmable DSS keys
- Blacklist: 100 Groups
- Call Log: 100 Logs
- Support 5 Remote Phonebook URLs
- LDAP
- Support Multiple Languages

IP-PBX Features

- SMS, Voicemail, MWI
- Music On-hold, Intercom, Multicast
- Call Pickup, Call Pickup in Group
- Anonymous Call
- Anonymous Call Rejection

Management

- Auto provisioning:
FTP/TFTP/HTTP/HTTPS/PnP
- Configuration via HTTP/HTTPS web
- SNMP, TR069
- Network capture
- NTP/Daylight saving time
- Firmware upgrade via web
- Syslog
- Configuration backup/restore
- Equipment Management System (NMS)

Audio

- HD Voice
- Wideband Codec: G.722, OPUS-WB
- Narrowband Codec: PCMA, PCMU, G.729, G723, G726, AMR, OPUS, iLBC
- Echo Cancellation/Dynamic Jitter
- Voice Activity Detection (VAD)
- Comfort Noise Generation (CNG)
- Full-duplex Speaker

Protocols & Standards

- SIP v1 (RFC2543), SIP v2 (RFC3261)
- SIP over TLS, SRTP
- TCP/IP/UDP
- RTP/RTCP, RFC2198, RFC1889
- HTTP/HTTPS/FTP/TFTP
- ARP/RARP/ICMP/NTP/DHCP
- DNS SRV/ A Query/NATPR Query
- STUN, Session Timer (RFC4028)
- DTMF: In-Band, RFC2833, SIP INFO
- LLDP
- OpenVPN, 802.1X

Physical Features

- Dimensions(approx.):
180x192x36mm(WxLxD) without foot stand
- Weight(approx.): device 370g
- Input voltage: 5Vdc
- Power supply: 5Vdc power adapter or PoE
- Power consumption: 1.4W
- Handset: 4P4C, RJ9
- Headset: 4P4C, RJ9
- Environmental conditions:
Operation temperature: 0~40°C
Humidity: 10% to 95% (non-condensing)
- 2xRJ45 10/100M Ethernet Ports
- Installation method: Desktop