

P10/P10P

Entry-level office IP phone

Flyingvoice P10/P10P Entry-Level office IP phone, with 132x64 pixels black and white LCD with backlight, providing users with a clear visual experience. P10/P10P supports 2 SIP accounts, integrated 100M dual network ports, PoE (P10P only) interface to meet the entry-level needs of corporate offices. It supports six conferencing, HD voice, and is easy to install and wall mount, meeting different application scenarios for desktop office and wall mount use. This easy-to-use, cost-effective IP phone can help you improve office efficiency and achieve quality voice communication.



All-new Design, Parctical details

P10/P10P adopts a new UI design, with a stylish and simple appearance and a friendly and easy-to-use operating interface, with a new breathing light caller alert, providing users with a more flexible and efficient voice collaboration experience.

HD Voice Quality, Better Calling Experience

P10/P10P uses advanced echo cancellation technology and intelligent noise cancellation technology to provide users with high-definition voice and achieve better voice calls.

Simple installation and easy deployment

P10/P10P with FRPS system and FACS platform for unified, fast and efficient bulk deployment, no need to change the existing IP network wiring, say goodbye to the complex deployment method, greatly reduce the time and labor costs, easy to manage, upgrade and maintain.

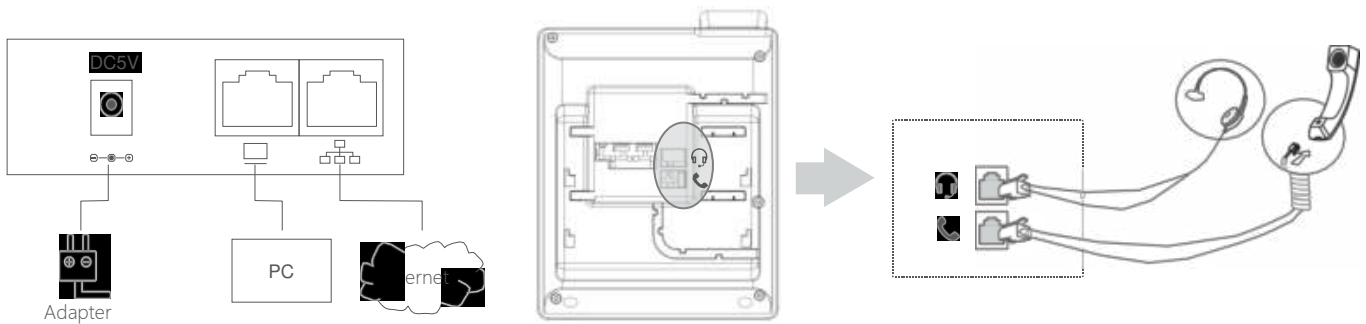
High Security, Strong Compatibility

P10/P10P are well protected by configuration files with AES, SRTP and TLS call encryption, etc. The P1X series IP phones are widely compatible with SIP based devices and PBX platforms such as Asterisk, Broadsoft and 3CX, etc.

- 128*64px backlit LCD screen
- 2 x 10/100 Mbps Ethernet port
- Supports PoE (P10P only)
- 6-Way conferencing
- 2 SIP accounts
- HD Voice
- RJ9 headset
- Wall mountable
- Opus

Phone Features	
LCD Screen	> 128*64px backlit B/W LCD screen > 16 bit depth color
Function Keys	> 12 Function Keys:home, directory, message, paging, call Hold, Record, mute, headset, transfer, redial, hands-free speakerphone > 4 Soft Keys, 4 Navigate Keys, Volume+/-
SIP Accounts	> 2 SIP accounts
Wi-Fi	> No supported
USB Port	> No supported
Ethernet Port	> 2*10/100M (adaptive) > Support PoE (P10P only)
Other Ports	> 1 x Handset port, 1 x RJ9 headset port
Call Features	
Basic function	> Call hold, mute, DND, redial, auto answer, One-touch speed dial, hotline, SMS, Call forward, call waiting, call transfer (attended/blind), Paging, Voice mail, BLF, Anonymous call, anonymous call rejection, Message waiting indicator(MWI), Hotdesk
Direct IP Call	> Supported
Phone book	> 2000 records; Phonebook search/import/export; XML/LDAP remote phonebook; Black list
Call history	> 60 records; Call history: dialed/received/missed
6-way conferencing	> Supported
Audio Features	
HD Voice	> HD handset, HD speaker
Wideband Codec	> G.722, Opus, AMR-WB(Optional), G.711WB(Optional), G.711(A/u), Ilbc, G.729A/B, G.723, G.726, AMR-NB(Optional)
DTMF	> In-band, RFC 2833, SIP info
Network	
Network Safety	> SIP Related Features: - SIP v1 (RFC2543), v2 (RFC3261),NAT Traversal by STUN, - SIP link mode: Proxy mode and peer-to-peer SIP link mode, - Call server redundancy supported > Network Features: - Static IP and DHCP, IPv4 /IPv6, Dual stack IPv4/IPv6, ARP/RARP, DNS A/DNS SRV/DNS NAPTR, - LLDP/CDP/DHCP VLAN; IEEE802.1X; OpenVPN; DSCP, 802.1p QoS > Time and Date synchronization: STNP, PBX, DHCP > Transport Feature: UDP/TCP/TLS; RTP/RTCP/SRTP
Configuration & Maintenance	
Auto-provision	> TR069 (TR104/TR106) > FTP/TFTP/HTTP/HTTPS for Auto Provisioning > Cloud device management: FACS > Cloud device management: FRPS > DHCP Option66

Configuration	<ul style="list-style-type: none"> > Supports Web configuration > Reset to factory, reboot > Package tracing export, system log
Others	
Physical Features	<ul style="list-style-type: none"> > New Stand, Wall mountable > AC/DC Adapter, AC Input: 100~240V, 50~60Hz > DC Output: 5V, 1A > Power consumption(PSU): 0.87~1.18W > Power consumption(PoE): 0.72~1.78W > Relative humidity: 10%~90% (No Condensing) - Storage temperature: -25~ 65°C - Operation temperature: 0~45°C > Dimension (W x D x H) : 200mm x 165mm x 47mm
Package Content	
Details	<p>Packing List:</p> <ul style="list-style-type: none"> > Flying Voice P10/P10P IP Phone、Power adapter (optional) > Handset with handset cord, Ethernet cable, Stand > Quick Start Guide <p>Size:</p> <ul style="list-style-type: none"> > Qty/CTN: 10 PCS、N.W/CTN: 7.5 kg、G.W/CTN: 9.47 kg > Giftbox size: 288×217mm×60mm > Carton size: 448mm×320mm×302mm
Certification & Compatibility	
Certificated by	
Compatible with	



About Flyingvoice

Flyingvoice is a leading manufacturer of VoIP phones, ATA& gateways and VoIP routers. These products are preferred globally by international carriers, ITSP's, ISP's, corporate companies, and home users alike.

Driven by a need for ongoing innovation within VoWLAN technologies, Flyingvoice is committed to manufacturing products that allow for ease-of-installation scenarios within an enterprise network. Our people continuously improve existing products as well as expand upon product applications tailored to clients' requirements. These improvements ensure that your experience provides for secure, effective and efficient collaboration – anytime & anywhere.

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Technical Support

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