

## D LINK IP PHONES

Connect the D LINK Broadband Internet IP Phones to your DSL/cable modem /router, and you can make phone calls anywhere in the world using the Internet. These IP phones gives you the advantages of VoIP, while retaining the same look and feel of a traditional advanced desktop telephone with the most cost efficient way.

### 150 SERIES SIP PHONE

#### PHONE FEATURES

- Multiuser (2 SIP accounts One for DSL/cable modem/remote router or Ethernet/Gigabit switch, the other port can connect to a PC
- POE Available
- Color ID display
- Built-in Speaker
- Cal History: 10 Missed Calls, 10 Received Calls, 10 Dialed Calls
- Phone book (up to 50 contact names and phone numbers)
- Supports XML Phonebook/Browser
- Supports click to dial via web Phonebook
- Day/Time display
- Cal/Time display
- 14 Selective Ring tones (9 tones & 5 melodies)
- 10 Sped dial number
- Incoming call indicator
- Password control for Configuration
- Pre-dial before sending



**DPH-150SE/B/F4 RP €79**

**SPECIAL PRICE € 70**

#### SIGNAL, MEDIA & NETWORK PROTOCOLS

- SIP RFC 3261 & the related RFC standard in Appendix A
- SDP RFC 2327
- RTP RFC 189
- IP assignment: Static IP, DHCP and PPOE
- STUN, static port mapping (for NAT traversal)
- SNTP
- DNS & DNS SRV
- TFTP/FTP/HTTP for Auto Provision
- IP/TCP/UDP/ARP/ICMP

#### QoS

- IEE 802.1p Compliant
- Diffserv (DSCP)/ToS

#### SECURITY/ CLASS OF SERVICE

- 802.1Q VLAN Tag + Full range VLAN

### DSS KEY EXPANSION MODULE

Optional for Receptions



**DPH-400EDM/B/F3 RP €49**

#### MODULE FEATURES

DPH-400EDM Expansion Module designed for D-link DPH-400SE/150SE Series IP Phones. It is an ideal solution for small businesses wanting to simplify incoming call handling by quickly directing calls to the requested party

The attendant console easily attaches to DPH-400SE Series IP Phone, providing 26 programmable speed-dial or direct station select (DSS) buttons, each illuminating the line's status (idle, ringing, busy, or null) via the BLF. Incoming calls can immediately be transferred to the proper location with the push of a button assigned to the extension.

The modular design of the DPH-400EDM permits up to 5 attendant consoles to be deployed, for a total of 130 buttons, offering an affordable solution that expands with the growth of your voice network. Installation and setup is just a matter of connecting the supplied bracket and cable to the attendant console, EDM uses external power supply.

**8 FXS PORT VOIP STATION GATEWAY**

**DEVICES INTERFACES**

- 8 FXS RJ-11 ports
- 1 10/100BASE-TX RJ-45 WAN port
- 4 10/100BASE-TX RJ-45 LAN ports

**WAN ACCESS TYPES**

- DHCP client
- PPPoE client
- Static IP address
- PPTP
- MAC cloning on WAN port

**DVG-5008S RPE €305**

**SPECIAL PRICE € 260**

**ROUTER**

- DHCP Server/ Virtual Server/ Port-triggered special application/DMZ/ Port filter
- IP filter/ MAC filter / URL filter/ RIP v.1, v.2/ Static Route/ DDNS/ NAT Traversal: pass-through NAT, STUN client, UPnP

**PHONE BOOK**

- Phone Book Manager Server (selected version), up to 200 IAD registrations
- Phone Book Manager Client

**NETWORK & FIREWALL**

- NTP (Network Time Protocol) support /DoS prevention engine (configurable)

**QoS**

- SIP and RTP always first policy/ WAN access rate control / LAN port rate limit, hi/low queue control

**SIP**

- By port registration / By IAD registration/ Outbound Proxy/Session timer: update, re-invite E.164 numbering/ Redundant SIP Proxy support (fail-over by registration)/ SIP PING (on special version) / Adjustable NAT port binding interval (binding with Proxy)/ CLIR (anonymous caller ID)/ Group hunting, configurable hunting sequence/ 3-way conference /T.30 fax pass-through T.38 fax relay

**MANAGEMENT**

- Web-based & Telnet/ By Provision Server/ 2 privilege WEB login accounts/STUN inquiry

**CALL FEATURES**

- Call Hold, Wait, Forward, Transfer
- Easy Tracking With Call Detailed Records
- Greeting Message, Caller ID Delivery

**D LINK VOIP ANALOG GATEWAYS**

*What This Product does*

The DVG5008S VoIP station gateways present an ideal Internet telephone solution for business use. These gateways convert voice traffic into data packets for transmission over the Internet, combine the industry's latest Voice over the IP network technology with advanced communication features, and are fully compatible with SIP Internet phone services. Designed as cost-efficient devices, they provide convenience and great savings for companies needing to place frequent long-distance and international business calls.



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**TECHNICAL SPECIFICATIONS**

**PROTOCOL STANDARDS**

- SIP (RFC 3261)
- SDP (RFC 2327)
- RTP (RFC 1889)
- RTCP (RFC 1889)
- Out-Of-Band DTMF (RFC 2833)
- IAX2 (RFC 5456)

**MANAGEMENT FEATURES**

- 30 extensions
- Supports 10 concurrent calls
- Single IP PBX supports multiple users across multiple sites
- Add external Analog Trunk Gateways to use standard phone-lines
- Save Money by using Internet Phone service (VoIP)
- User-Friendly Administration Interface
- Web-based Monitoring and Administration
- Call Statistics and
- Call Detail Records (CDR)
- Support HTTP upload
- Add / delete the IP phones batchRing Group recordTrunk backup
- Auto config IP phone in a LAN network

**CALLING FEATURES**

- Business Calling Features
- Caller ID
- Call Transfer (Blind Transfer and Assisted Transfer)/Call History/Call Hold
- Call Forwarding (Always/on Busy/on No Answer/Follow me)
- Call Park
- Ring Group
- Call Pickup
- Video Calls
- VPN Client ((PTP/L2TP)
- VPN Server (PTP/L2TP)
- DDNS Client (Support Dyndns.org)
- Support SKYPE for SIP

**IVR/AUTO-ATTENDANT FEATURES**

- Music on Hold
- Configurable IVR Menu
- Voice Mail
- Mailbox Access control (PIN)
- Configurable Mail Box
- Notification via email

**SMB IP PBX**

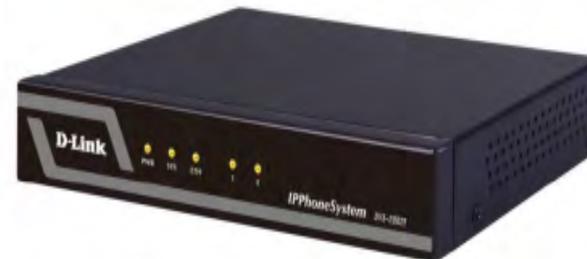
**HARDWARE**

- Processor: 128MB DR2 embedded RISC DSP
- RAM: 4GB, 128M Onboard SDRA<sup>1</sup>
- USB: 1 external USB 2.0 port

**CONFIGURATION**

- Secure Web Based Management
- Configuration Backup/Restore
- Asterisk API based

**SAVE COST!**  
 Save money by using Internet Phone Service, integrates well with known ITSPs, connect to your office on the Go



**DVX-2002F RP €355**



**SPECIAL PRICE €315**

*What This Product does*

The all-in-one DVX 2002F IP PBX can not only provide the traditional basic PBX features (call hold, call forwarding, call waiting, video call, etc.), but also provide enhanced features such as visual operator, voice mail to mail, multi-media music on hold, and auto attendant, etc. In addition, it's very convenient for SMEs' management and maintenance, also easy to upgrade. SMEs can set up own phone system to improve the company image and office efficiency.

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**TECHNICAL SPECIFICATIONS**

**PROTOCOL STANDARDS**

- SIP (RFC 3261)
- SDP (RFC 2327)
- RTP (RFC 1889)
- RTCP (RFC 1889)
- Out-Of-Band DTMF (RFC 2833)
- IAX2 (RFC 5456)

**MANAGEMENT FEATURES**

- 300 extensions
- Supports 60 concurrent calls
- Single IP PBX supports multiple users across multiple sites
- Add external Analog Trunk Gateways to use standard phone-lines
- Save Money by using Internet Phone service (VoIP)
- User-Friendly Administration Interface
- Web-based Monitoring and Administration
- Call Statistics and
- Call Detail Records (CDR)
- Support HTTP upload
- Add / delete the IP phones batch Ring Group record Trunk backup
- Auto config IP phone in a LAN network

**CALLING FEATURES**

- Business Calling Features
- Caller ID
- Call Transfer (Blind Transfer and Assisted Transfer)/Call History/Call Hold
- Call Forwarding (Always/on Busy/on No Answer/Follow me)
- Call Park
- Ring Group
- Call Pickup
- Video Calls
- VPN Client (Support N2N)
- DDNS Client (Support Dyn dns.org)
- Support SKYPE for SIP

**IVR/AUTO-ATTENDANT FEATURES**

- Music on Hold
- Configurable IVR Menu
- Voice Mail
- Mailbox Access control (PIN)
- Configurable Mail Box
- Notification via email

**SMB Enhanced IP-PBX**

**CONFERENCE SERVER**

- Access Control
- Conference Recording
- Conferencing from external lines

**HARDWARE**

- Processor: Intel intel Atom 1.8 ghz
- RAM: 2GB
- USB: 2 external USB 2.0 port
- Hard disk: 250 GB 2.5" **INCLUDING**

*-Includes DVX-8010 Expansion module for maximum 4 modular interface*

**CONFIGURATION**

- Secure Web Based Management (Elastix GUI and FreePBX)
- Configuration Backup/Restore
- Asterisk API based

**SAVE COST!**  
 Save money by using Internet Phone Service, integrates well with known ITSPs, connect to your office on the Go

**DVX-8000 RP €1,120**

**SPECIAL PRICE € 960**



Optional :DVX FXO 8 Card



<p><b>DVX-8020</b></p> 	<p><b>DLINK ASTERISK FXO MODULE 8-PORT FOR DVX-8000 INCLUDING ONE TIME LICENSES ACTIVATION FEES</b></p>	<p><b>DP €319</b></p> <p><b>SPECIAL PRICE € 265</b></p>
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*What This Product does*

The DVX-8000 offers all of the essential telephony features required for medium to enterprise businesses. It can utilize standard phone lines via an external phone line gateway or cost effective Internet telephony services. The device can be equipped with up to 32 analog ports, or a single PRI module. The DVX-8000 supports up to 160 PSTN Analog phones ports with external Gateway units. It can handle up to 300 users and up to 60 concurrent calls.

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