

DVX-9000 COMMUNICATION - ANYWHERE, ANYTIME



DVX-9000

xStack IP Telephony

IP PBX Asterisk[®] Based

SAVE COST

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

MODULAR & SCALABLE

Expand your existing analog telephony infrastructure, supports modular analog and digital interface & features SIP/IAX trunks.

SIMPLE INTEGRATION

Provides user-friendly administration interface and is flexible to integrate with third-party software - Asterisk API

IP TELEPHONY

The DVX-9000 is an Intel-Core2Duo based Asterisk IP PBX. The DVX-9000 features Elastix Asterisk distribution with various combinations of telephony ports. The device can be equipped with up to 32 analog ports, 4 E1 PRI. The DVX-9000 supports up to 800 PSTN Analog phones ports with external Gateway units. It can handle up to 1000 users and up to 300 concurrent calls.

Internet IP telephony, also called Voice over IP (VoIP), is defined as the transport of telephone calls over the Internet as standard Internet data packets. Internet telephone calls can originate from traditional phone handsets via phone line-to-Internet (Analog Trunk) gateways, by PCs using software, or embedded devices (IP Phones). Most of the interest in Internet telephony is motivated by cost savings and ease of developing and integrating new services. Internet telephony integrates a variety of services provided by the current Internet and the Public Switched Telephone Network (PSTN) infrastructure.

The DVX-9000 offers all of the essential telephony features required for medium to enterprise businesses. Features such as call forwarding, call hold, follow me, and voice mail. Incoming calls are directed by the integrated auto-attendant and hunt groups to assist callers to their destinations. It can utilize standard phone lines via an external phone line gateway or cost effective Internet telephony services.

EXTENSIONS ANYWHERE

The DVX-9000 supports up to 1000 extensions, which can be located anywhere with Internet access. Multiple units can be used to increase the number of extensions or unite a company that has many locations under a single PBX system.

EASY WEB CONFIGURATION

The PBX phone features are user adjustable via the DVX-9000's web configuration tool. The administrator assigns each extension a profile of telephony features, which allows the best match for a user's job function. Each user can fine-tune their assigned profile via the web to match their daily business schedule.

CONFERENCE BRIDGE

Phone conferencing is typically an expensive external hardware or service. The DVX-9000 includes a phone conferencing bridge, which makes it unsurpassed for value and features. Users are able to schedule and invite parties to conferences via the web configuration.





KEY BENEFITS OF THE DVX9000

AS PBX:

- Configurable as core IP or hybrid PBX.
- Switches calls & Manages routes.
- Connects callers with the outside world over IP/analog (POTS) and digital connections.

AS GATEWAY:

- Configurable as media gateway.
 Bridges legacy PSTN to the expanding world of IP telephony.
- Conversion between a wide range of communications protocols and media codecs.

AS MEDIA/FEATURE SERVER:

- Provides IVR and Conference Bridge.
 Automated attendant and unified
- messaging.
- Can replace aging legacy voicemail systems.

IN CALL CENTER:

- Features built-in ACD systems.
- Additional remote IP agent capabilities.
- Advanced skills-based routing.

| D-Link MEA Office | Telephone No.: |
|-------------------|---|
| D-Link UAE | +971-4-880-9022 |
| D-Link Egypt | +2-02-2671-8375 +2-02-2671-7280 |
| D-Link Morocco | +212-663-727-324 |
| D-Link KSA | +966-1-217-0008 |
| D-Link Nigeria | +234-1-8536-769 |
| D-Link Pakistan | +92-51-2800-397-98 +92-21-454-8158, +92-21-454-8310 |
| D-Link Kenva | +254-20-434-3202/3/4 |

PROTOCOL STANDARDS

- 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

- 1000 extensions
- Supports 300 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)
- CALLING FEATURES Business Calling Features
- = Caller ID
- Call Transfer (Blind Transfer and Assisted Transfer)
- Call History
- Call Hold
- Do Not Disturb
- Call Forwarding (Always/ on Busy/on No Answer/ Follow me)
- Call Park
- Ring Group
- Call Pickup

IVR/AUTO-ATTENDANT FEATURES

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

CONFERENCE SERVER

- Access Control
- Conference Recording
- Conferencing from external lines

HARDWARE

- Processor: Intel Core2Duo
- RAM: 1GB
 Hard disk: 250 GB(500GB
- Optional)
- USB: 2 external USB 2.0 ports

TECHNICAL SPECIFICATIONS

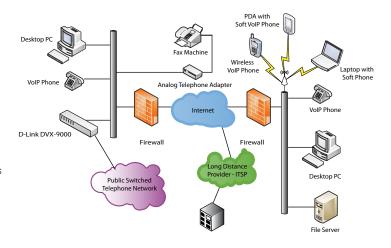
CONFIGURATION

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

| EXPANSIO | EXPANSION MODULES & INTERFACE CARDS | | |
|----------|---|---------|--|
| DVX-8010 | Expansion module for DVX-9000 for maximum 4 modular interface | AASTA | |
| DVX-8020 | 8 port FXO module for DVX-9000 | | |
| DVX-8030 | PRI module for DVX-9000 | CRIIFIC | |

| VOIP GATEWAYS | | |
|---------------|---|--|
| DVG-2102S | 2 port Analog VoIP Telephone Adapter (ATA), 1WAN, 2FXS, with PPTP, L2TP VPN support | |
| DVG-5004S | VoIP Gateway with built-in 4 FXS, 1 10/100Mbps WAN & 4-port 10/100Mbps Switch | |
| DVG-5008S | VoIP Gateway with built-in 8 FXS, 1 10/100Mbps WAN & 4-port 10/100Mbps Switch | |
| DVG-6004S | VoIP Gateway with built-in 4 FXO, 1 10/100Mbps WAN & 4-port 10/100Mbps Switch | |
| DVG-6008S | VoIP Gateway with built-in 8 FXO, 1 10/100Mbps WAN & 4-port 10/100Mbps Switch | |
| DVX- 8050 | 2 Port USB External Device for DVX9000 redundancy Twining | |

| PHYSICAL ENVIRONMENT | | |
|----------------------|---|--|
| Power Inp | ut | |
| Power sup | ply | Internal |
| Voltage | | Switching, auto-adjust 110/120 Volts, 50/60z |
| Power con | sumption | 230 Watts (maximum) |
| Built-in gro | ounding connection | A. |
| Environment | | |
| Storage te | mperature | -20°to 70°Celsius (-4°- 158°F) |
| Working te | mperature | 0° to 40° Celsius (32 $^{\circ}$ - 104 $^{\circ}$ F) |
| Humidity | | 20%-95%, non condensing |
| Dimensions & Weight | | |
| Size | 19″ 2U industry standard rack-mountable chassis | |
| Weight | 8 Kg 18Lbs (weight depends on configuration) | |



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IP TELEPHONY