

Product Specification

Internet Telephony PBX System

IPX-300

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Change History:

Revision	Date	Author	Change List
Version 1.0	October 31, 2007	Brian Chen	Initial release

Author	Brian Chen	Editor:	Brian Chen
Reviewed by:	Jonas Yang	Approved by:	Tom Shih



1. PRODUCT DESCRIPTION

PLANET IPX-300 IP PBX telephony systems are designed and optimized for the small business in daily communications. It can support up to 30 user registrations and easy to install and manage a fully working system with the convenience and cost advantages. The future IP PBX telephony system offers all of the essential features of telephony which is required by small business users for their telecommunication/data needs.

The IPX-300 is a feature-rich SIP based IP PBX telephony system that integrates NAT functions to make it perfect for small business usage. The IPX-300 integrates traditional PBX system functions and provides many advanced functions including voice mail to email, web management etc. Designed to run on a variety of VoIP applications, the IPX-300 provides IP-based communications, voice conferencing, call detailed record (CDR), centralized Auto-Attendant (AA), and Interactive Voice Responses (IVR). The IPX-300 utilizes standard PSTN /GSM lines via the interfaces of FXO/GMS gateway to become a feature-rich IP PBX telephony system that supports seamless communications among existing local calls, SIP-based endpoints including low cost of long distance service, telephone number portability and one network for both voice and data.

With the IPX-300, standard SIP phones can be easily integrated in user's office. Users may integrate PLANET IP Phone VIP-154 series, VIP-155PT/ 350PT/ 550PT, the VIP-156/ 157/ 158/ 161SW of ATA (analog telephone adapter) series, and Gateway series VIP-281/ 281GS/ 480 to build up the VoIP network deployment in minutes.

2. PRODUCT FEATURES

PBX Features

Automated Attendant (AA) Interactive Voice Responses (IVR) Voicemail support Call Detailed Record User Management via Web Browsers Call / Pickup Group Display 30 Registered User's Status: Unregistered / Registered / On-Call

Call Features

Call Forward Immediate Call Forward on Busy Call Forward on No Answer Call Pickup / Call Park Caller ID Music on Hold / Music on Transfer Call Transfer / Call Hold / Call Waiting Three-way conference with feature phones (VIP-154 series, VIP-155PT/ 350PT/ 550PT and VIP-156/ 157/ 161SW series)

Router/Firewall Features

DHCP Server for LAN Users Access Control / URL Filter Virtual Server / DMZ/ Port Mapping Static Route Pass-through UPnP

3. PRODUCT SPECIFICATION

3.1 MAIN COMPONENT

CPU:	ARM926EJS
RAM:	32M SDRAM CT51V561601A
FLASH:	8M Flash MX29LV640D
Switch:	IC+ IP101A



	1		
LAN	1 (10Base-T/100Base-TX, Auto-Negotiation)		
WAN	1 (10Base-T/100Base-TX, Auto-Negotiation)		
Protocols and	IEEE 802.3 (Ethernet)		
Standards	IEEE 802.3u (Fast Ethernet)		
	RFC 768 UDP		
	RFC 783 TFTP		
	RFC 791 IP		
	RFC 792 ICMP		
	RFC 793 TCP		
	RFC 826 ARP		
	RFC 1034, 1035 DNS		
	RFC 2136 DDNS		
	RFC 1631 NAT		
	RFC 2068 HTTP		
	RFC 2131 DHCP		
	RFC 2516 PPPoE		
	RFC 3261, RFC 3311, RFC 3515		
	RFC 3265, RFC 3892, RFC 3361		
	RFC 3842, RFC 3389, RFC 3489		
	RFC 3428, RFC 2327, RFC 2833		
	RFC 2976, RFC 3263		
Cabling	10Base-T - 2-pair UTP Cat. 3, 4, 5, up to 100m (328ft)		
U U	100Base-TX – 2-pair UTP Cat.5, up to 100m (328ft)		
Management	HTTP Web Browser		
Internet Connection	Fixed ID DDDoE DUCD alignet		
Туре	Fixed IP, PPPoE, DHCP client		
PBX Feature	Automated Attendant (AA)		
	Interactive Voice Response (IVR)		
	Voicemail support		
	Call Detailed Record		
	User Management via Web Browser		
	Call / Pickup Group		
	Display 30 Registered User's Status: Unregistered / Registered / On-Call		
	Call Forward Immediate		
	Call Forward on Busy		
	Call Forward on No Answer		
Call features	Call Pickup / Call Park		
	Caller ID Music on Hold / Music on Transfer		
	Call Transfer / Call Hold / Call Waiting		
	Three-way conference with feature phones		
	(VIP-154 series, VIP-155PT/ 350PT/ 550PT and VIP-156/ 157/ 161SW series)		
Voice CODEC	G.711 a/u, G.723.1 (5.3, 6.3kbps), G.729A (8kbps), GSM, ILBC		
Voice processing	DTMF detection/generation, Support password authentication using MD5 digest		

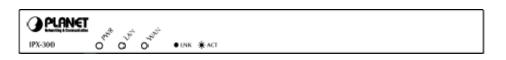
3.2 PHYSICAL SPECIFICATION

Dimensions

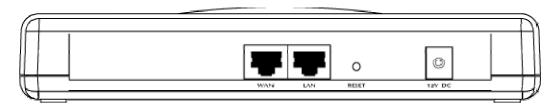
180 x 110 x 25 mm (W x D x H)



Front Panel



Rear Panel



LED definitions

Front Panel LED	State	Descriptions
PWR	On	PBX Power ON
FWN	Off	PBX Power OFF
	On	PBX network connection established
WAN Port	Flashing	Data traffic on cable network
	Off	Waiting for network connection
	On	LAN is connected successfully
LAN Port	Flashing	Data is transmitting
	Off	Ethernet not connected to PC

3.3 ENVIRONMENTAL SPECIFICATION

Operating

Temperature: 0~40°C Relative Humidity: 10~90% (non-condensing)

3.4 ELECTRICAL SPECIFICATION

Input Voltage: 12V DC, 500 mA Power Consumption: 6 watts

3.5 REGULATORY COMPLIANCE

FCC Part 15 Class B CE Class B

3.6 RELIABILITY

MTBF > 50,000Hrs @25degree C MIL-STD-781C



3.7 BASIC PACKAGING

- Internet Telephony PBX system unit
- Power Adapter
- Quick Installation Guide
- User's Manual CD

3.8 PACKAGE INFORMATION

- Gift Box Dimension 305 x 205 x 67 mm (W x D x H)
- Gift Box Weight 0.72 Kg (gross weight)
- Carton Dimension 750 x 630 x 440 mm (W x D x H)
- Carton Weight 15.94 Kg (gross weight)

Carton Units

20 units