

IPNext50 Small, Powerful IP-PBX

Next Generation IP-PBX System for the Small Business and Home Office

Product Overview





AddPac Technology

Sales and Marketing

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IPNext50 Next Generation IP-PBX

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AddPac

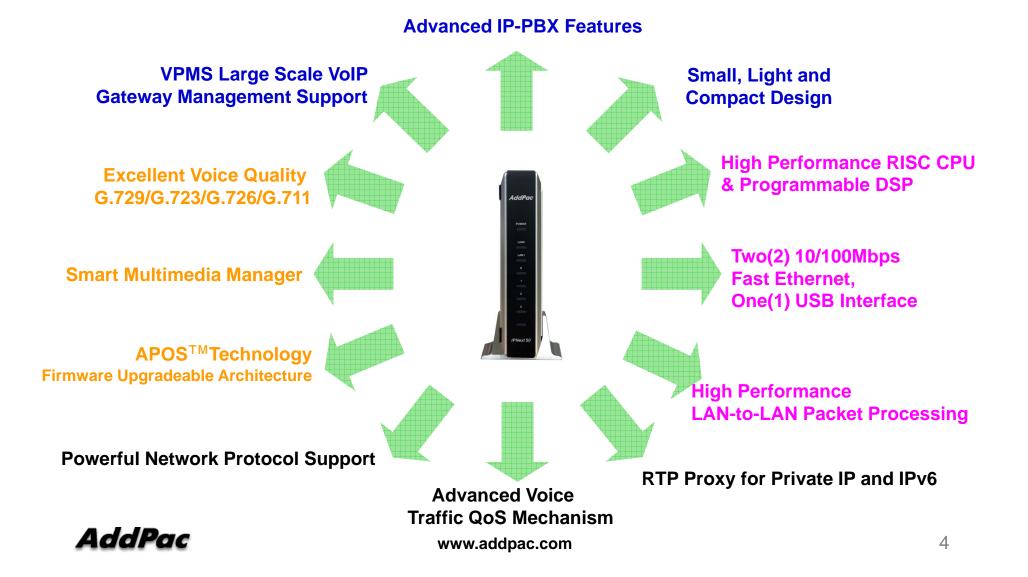
Product Overview

IPNext50 Next Generation IP-PBX

- SIP Application Server, Proxy, Registrar and Location Server
- Multiple ITSP Trunk with SIP & H.323 Accounts Support
- High Performance RISC & Programmable DSP Architecture
- Two(2) 10/100Mbps Fast Ethernet (IP Share ,etc)
- One(1) USB Port for USB Memory and Wireless LAN
- Up to Four(4) PSTN FXO VoIP Interface
- High Performance LAN-to-LAN Routing Capability
- VoIP Gateway : G.711/G.726/G.723/G.729, T.38 Fax , VAD, etc
- Powerful Network Protocols (PPPoE, DHCP, Static Routing, etc)
- IPv4/IPv6 Dual Stack
- RTP Proxy Function Embedded for Private IP and IPv6 Address Interworking
- Presence Service Features for Smart Multimedia Messenger and Smart IP Phone
- IVR Scenario Editor, Voice Mail, Third Party Conference (G.711)
- Firmware Upgradeable Architecture
- Smart Multimedia Manager for IP-PBX Management
- Smart Messenger Service for UC
- VPMS (VoIP Plug&Play Management System) for Large Scale Deployment
- Advanced Voice QoS Mechanism
- Small, Light and Compact Design
- Power Switch for Stability and Status LEDs

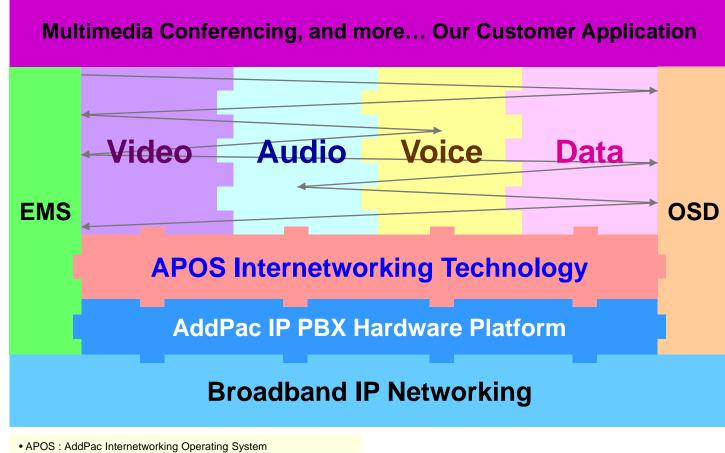
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Product Highlights



APOS Technology

IPNext50 Next Generation IP-PBX



OSD : On- Screen Display

EMS : Element Management System



Hardware Specification

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- **RISC Microprocessor Computing Power**
- High End Programmable DSP
- VoIP Gateway Interface
 - IPNext50 Model A: Basic Configuration
 - IPNext50 Model B: Two(2) FXO Port
- Network Interface
 - Two(2) 10/100Mbps Fast Ethernet (RJ45)
- USB Interface
 - One(1) USB Host Mode Interface for USB Memory and Wireless LAN
- Power Supply
 - External Power Adaptor
- Power On/Off Switch
- Power LED, LAN LEDs, FXO Port LEDs



RISC

CPU

High-end

DSP

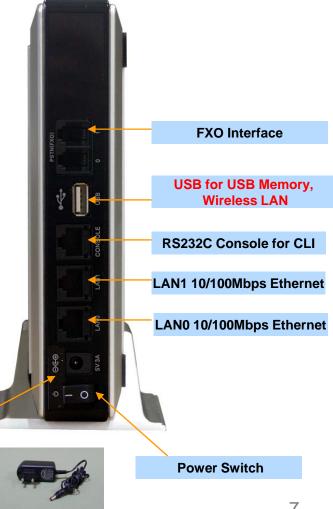
Hardware Specification

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Hardware Specifications

IPNext 50 Series	Basic Specifications
CPU	High End RISC Microprocessor
Voice Interface (Option)	2 FXO Voice Interface(RJ-11) :Model B
Ethernet Interface	2-Ports 10/100Mbps Ethernet Interface(RJ-45)
Flash Memory	High-speed Flash Memory for System, Voice Mail, etc
System Main Memory	128Mbyte High-speed SDRAM
USB Interface	USB Host Mode Interface for USB Memory, Wireless LAN, USB Hub.
Power Requirement	External Power Supply Adaptor / VAC 110~220V, 50/60Hz, 5V 3A
Operating Temperature	0°C ~ 45°C (32 °F ~ 122°F)
Storage Temperature	-40°C ~ 85°C (-40°C ~ 185°F)
Relative Humidity	5% ~ 95% (Non-condensing)
Dimensions	182mm x 182mm x 38mm (W x D x H)

Network interface Configurations



External Power Interface

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Signaling Server

- SIP Application Server, Proxy, Registrar and Location Server (RFC3261)
- Multiple ITSP Trunk with SIP & H.323 Accounts Support
 - IP UA Client Role for Registering to ITSP SIP Server
 - H.323 Gatekeeper Client Role for Registering to ITSP H.323 Gatekeeper Server

• IVR & Auto Attendant

- Default Auto Attendant Support
- Interactive Voice Response (IVR)
 - Provides with GUI-based Smart IVR Scenario Editor
 - Upload/Download Scenario by Smart IVR Scenario Editor
 - Supports Multiple Concurrent Scenarios
 - Supports Recordable IVR Prompts

Voice Mail

- Support Voice Mail with IVR
- Access from Remote Site via Trunk Support
- Voice Mail Notification Support



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• Conference

- G.711 u-law, G.711 a-law Internal 3-Party Audio Conference Support
- Ad-hoc Conference
- Dial-Out Conference
- Meet-me Conference
- Multiple External MCU support(Video, Audio, etc) : AddPac AP-MC1000, etc
- Conference Chair and Participants Management

Music & Announcement

- Music on Hold
- Replaceable Announcements
- Dialing Music / Tone Service



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• Number & Call Routing

- Trunk Hunting by Preference or Sequential
- Call Hunting by Preference, Simultaneous, Random
- Call Hunting by Chained Hunting Group
- Partition for Address Grading
- Call Class for Call Access Control
- Number Translation Rule for Inbound/Outbound Call
- Centrex with Prefix Support
- Multiple Shared Devices with One Number
- Multiple Numbers on One Device
- Individual Call Park within Park Number Pool
- Group Call Park within a Group or Other Group
- Call Pickup of Ringing Call of Same Group or Other Group
- Call Pickup of Parked Call
- Call Transfer Blind, Consult
- Call Forwarding Unconditional, Busy, No Answer, Voice Mail
- Call Waiting
- Call Swapping
- Call Hold





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• IP-PBX Advanced Features with AddPac IP, Video Phones

- Multiple Call Handling with Call Status and Calling Line Number and Name
- Plug and Play with Auto Discovery Function
- Softkey Map Download and Control
- Time and Date Setting
- Voice Mail List View
- Parked Call List View
- Call Forward Setting
- Recent Call List View
- Calling Number and Name Identification
- Individual Call Park within Park Number Pool by Softkey
- Group Call Park within a Group or Other Group by Softkey
- Call Pickup of Ringing Call of Same Group or Other Group by Softkey
- Call Pickup of Parked Call by Softkey
- Call Transfer Blind, Consult by Softkey
- Call Waiting Indication
- Call Swapping by Softkey
- Call Hold by SoftKey
- Conference Control



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User & Device Management

- LDAP (Lightweight Directory Access Protocol) Support
 - Supports Hierarchical Organization
- Auto Discovery of IP Phones & Video Phones
- Monitoring Status of Phones
- Miscellaneous*
 - Distinctive Ring by Calling User
 - Auto Config & Upgrade
 - Intercom
 - Personal Directory
 - Downloadable Ring
 - Do not Disturb



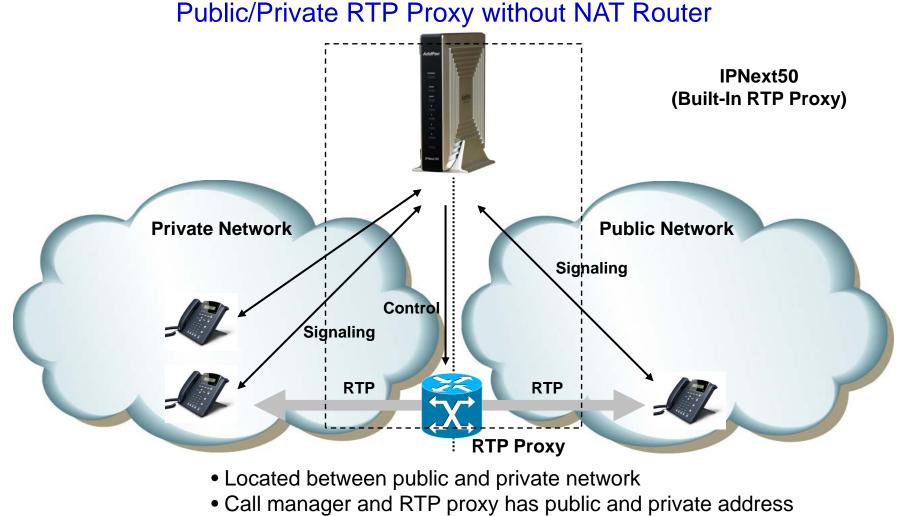


RTP Proxy Function

- Protocol Independence (H.323, SIP)
- Built-In RTP Proxy for Small Office
- Support RTP Proxy without NAT
- Support Symmetric / Asymmetric NAT
- Support IPv4/IPv6 RTP Proxy
- Support RTP Broadcast
- Support RTP Conference



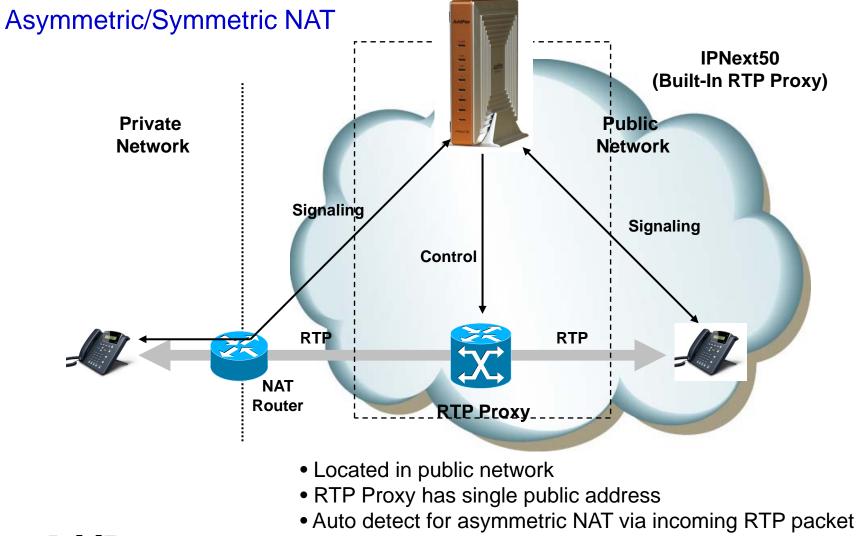
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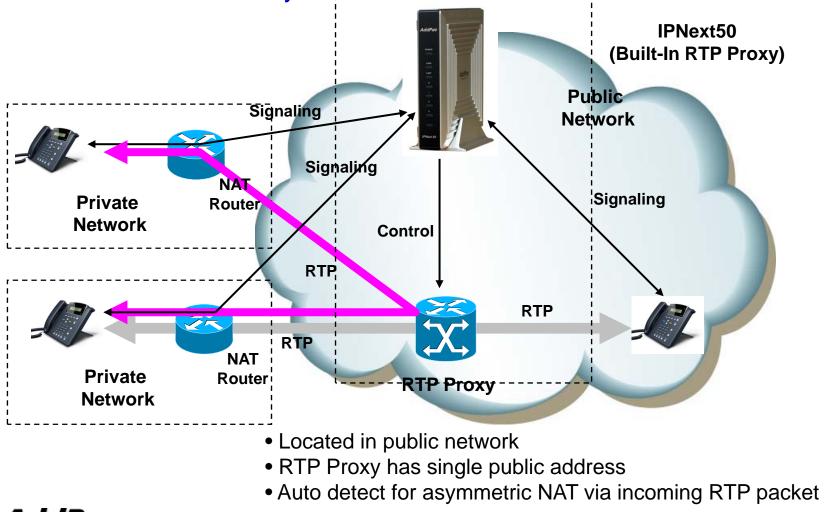
• Call manager determine that the call is internal or external www.addpac.com

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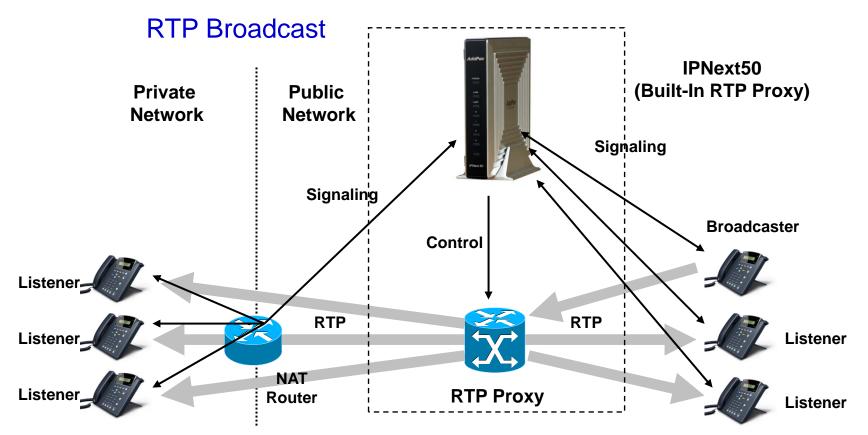
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RTP Proxy Communication between Private IP



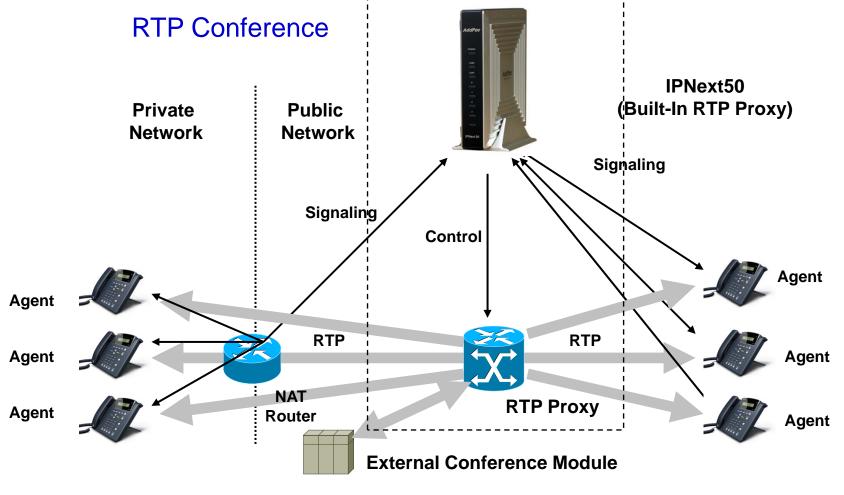
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- Call Manager set the broadcaster and listener via control message
- One broadcaster and multiple listener
- RTP proxy relay for broadcaster RTP traffic
- RTP proxy never relay for listener traffic



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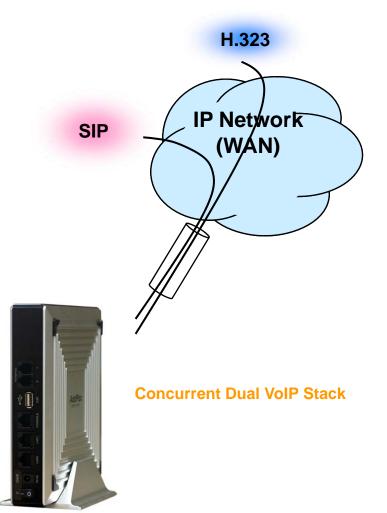


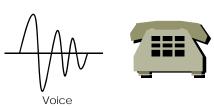
• RTP Proxy communicate with external/internal conference module (mixer)

Conference support for private address, asymmetric NAT



- H.323, SIP Dual VoIP Signaling Stack
- H.323
 - ITU-T Standard H.323 v3 Support
 - Support H.245 Tunneling
 - Including H.235 Security Features
- SIP
 - IETF RFC3261 or RFC2543 SIP Standard







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• H.323

- Fast connect, normal connect support
- H.245 tunneling support
- Q.931 response message setting for inbound VoIP calls
- H.245 logical channel open timing selection function
- Start H.245 procedure support
- DTMF / Hook flash relay with H.245 alphanumeric / signal
- Secondary gatekeeper support
- Gatekeeper assignment according to the domain name
- Gatekeeper discovery with multicast
- Lightweight RRQ support
- Signaling TCP port assignment
- Resource threshold setting with RAI
- H.235 clear-token, crypto-token support
- canMapAlias support
- Technical prefix (supported prefix) support
- Public IP assignment in NAT environment

• SIP

- Gateway-based / Endpoint-based registration support
- Secondary proxy-server assignment function
- SIP signaling port change function
- SIP proxy server assignment according to the domain name
- T.38 real-time fax relay support
- DTMF relay support with RFC2833 / OPTION message
- Re-INVITE support





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- Voice Codec
 - G.711 A-Law, G.711 U-Law
 - G.726 r16, G.726 r32
 - G.729A
 - G.723.1 r63, G.723.1 r53
 - VAD (Voice Activity Detection) function support
 - DTMF relay support (H.323, SIP, MGCP common) based on RFC2833

• RTP

- Redundant RTP packet transmission in case of severe packet loss
- Dynamic jitter buffer management and RPT packet jitter and loss compensation with heuristic & DSP error concealment
- Static jitter buffer setting support
- Voice frame per RTP packet number control for each codec
- In-band ring-back tone support
- Virtual ring-back tone support
- Tone parameter change support

• FAX

VoIP

- Fax relay mode supporting T.38, inband-T.38, bypass mode
- Lost packet compensation with redundant setting in case of T.38 fax relay
- Fax relay mode, rate setting for remote end



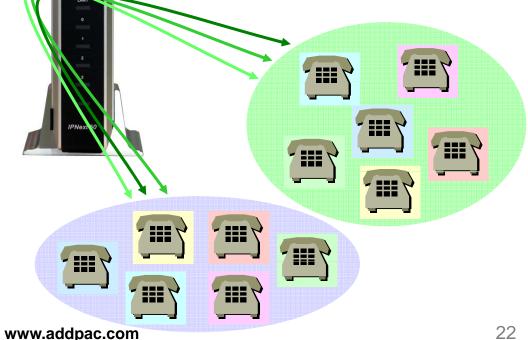
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VoIP Call Controls

- Hot line connection function with PLAR (Private Line Auto Ring Down)
- Leased line emulation function _
- Connection monitoring function
- Fault tolerant with Redundancy and Call Distribution among Gateways for load balancing
- Call attempt with IP address
- H.323, SIP, MGCP inbound call connection for each voice port
- Multiple E.164 setting for one voice port
- One E.164 or digit pattern can be assigned to more than one voice port
- Hunting with Longest match/ priority/ sequence/ random
- One stage call setup by Digit forwarding _
- Call barring with specific digit patterns
- Calling and called number conversion for PSTN outbound calls
- PSTN rerouting in case of VoIP call attempt failure

VoIP Call Controls (cont.)

- Call transfer for internal calls
- Call pickup for internal calls
- Calling and called number conversion for VoIP outbound calls
- Calling and called number conversion for VoIP inbound calls
- Fax broadcasting call control





AddPac

POWER

Advanced QoS Features

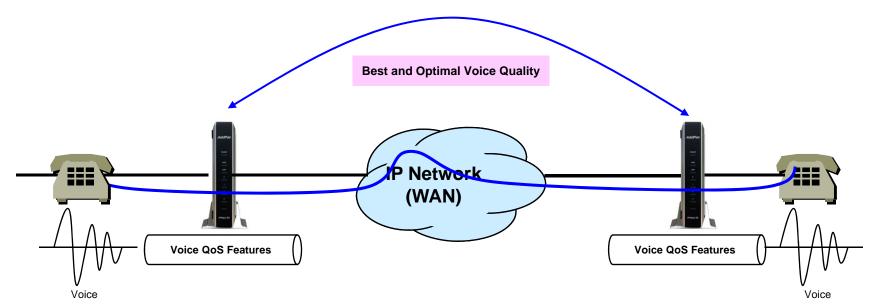
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• Enhances Transmit Voice QoS Features

- Voice Traffic Priority Queuing
- QoS Service Profiling
- Providing Virtual Network Transmit Algorithm
- Real-time Voice Traffic QoS Support
- RTP Packet Transmit Interval Control
- Supporting RTP Packet Redundancy Scheme
- IP Header Control such as ToS, Diffserv

• Enhances Receive Voice QoS Features

- Dynamic Jitter Buffer Management
- Error Concealment
- Support T.38 FAX Data Error Recovery Scheme





Network Protocols (Cont.)

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Basic Network Protocols

- ARP, IPv4, TCP, UDP, ICMP, SCTP, IGMP, MLD

Routing Protocol

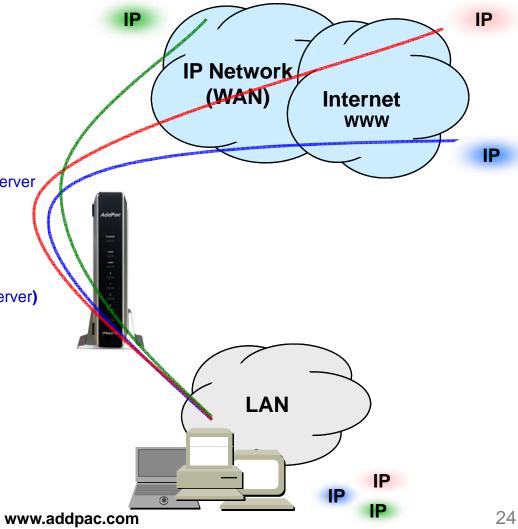
- IPv4 : Static
- IPv6 : Static

Service Protocol

- FTP, Telnet, TFTP, DHCP Server/Relay, SNMP Server
- CDP (Cisco Discovery Protocol)
- DNS Resolver , DDNS(nsupdate)
- Bridge
- Syslog
- IP/IPv6 policy control (QoS)
- VPDN (Virtual Provate Dial-up Network : L2TP Server)

Miscellaneous

- Cisco Style CLI
- Standard & Extended IPv4 Access List
- Multi-level User Account Management
- IP accounting
- STUN Client





Network Protocols

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IPv4/IPv6 Interworking

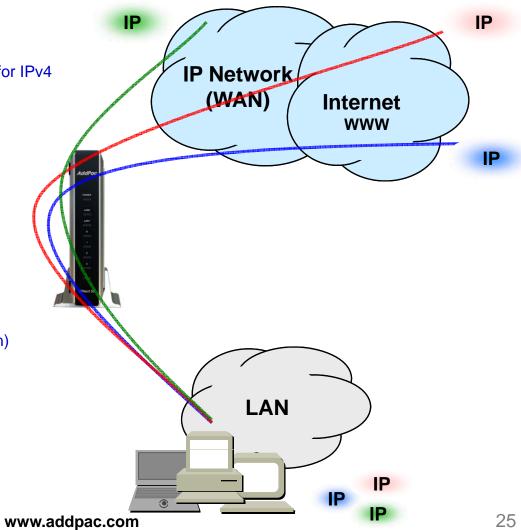
- NAT/PAT for IPv4
- IP connect (formerly ip-share) and device cascade for IPv4
- IP/IP, IP/GRE tunneling
- NAT-PT
- 6to4, Autoconfig tunneling

IPv4 Address Configuration

- Fixed (Static)
- DHCP
- PPPoE

IPv6 Address Configuration

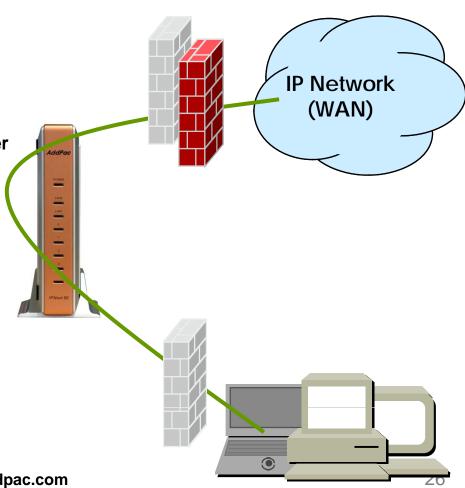
- Fixed (Static)
- EUI-64
- Autoconfig (Neighbor Advertisement and Solicitation)





Security Management

- IP packet filtering
- IP access list
- User authentication function
 - Password Authentication Protocol (PAP)
 - Challenge Handshake Authentication Protocol (CHAP)
- Enable/Disable specific protocols
- Auto-square connect of Telnet session
- Account Management function for multi-level user
- SNMP/TELNET/FTP/HTTP/TFTP port assignment function
- SNMP/TELNET/FTP access list management
- Boot mode security checking function





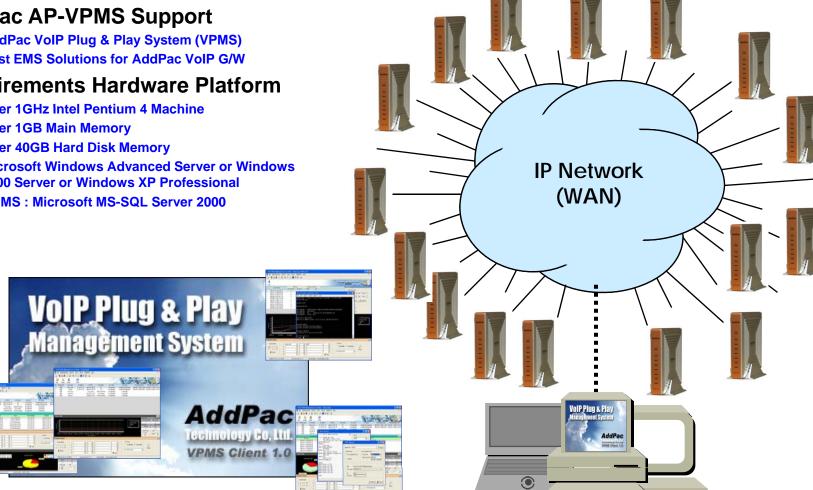
Smart Multimedia Manager

- Web based Smart and Easy Management Tool Support
- User, Device, Call, Conference Management with Intuitive User Interface
- Built in Event Manager or External Event Manager Support
 - Event Configuration with Event Source Filter and Event Logging Filter
 - Event Monitoring with Colorful Format
 - Debug Level Syslog Monitoring Support
 - Event Analysis by Smart Event Analyzer
- Active Call Monitoring Support
- Call Tracing Support
- Active Conference Monitoring Support
- Call History Viewer Support
- Backup, Restore and Initializing of Database Support
- Export and Import of Database to Excel File Support



VPMS(VoIP Plug&Play Management System)

- AddPac AP-VPMS Support
 - AddPac VoIP Plug & Play System (VPMS)
 - Best EMS Solutions for AddPac VoIP G/W
- Requirements Hardware Platform
 - Over 1GHz Intel Pentium 4 Machine
 - Over 1GB Main Memory
 - Over 40GB Hard Disk Memory
 - Microsoft Windows Advanced Server or Windows 2000 Server or Windows XP Professional
 - DBMS : Microsoft MS-SQL Server 2000





IP Multimedia Terminals

- AP-VP500 Video Phone
- AP-VP350 MCU Video Phone
- AP-VP300 Video Phone
- AP-VP280 Video Phone
- AP-VP150 Video Phone
- AP-VP120 Video Phone
- AP-IP300 Premium IP Phone
- AP-IP160 IP Phone
- AP-IP120 IP Phone
- AP-WP100 WiFi Phone
- AP-SMP100 Soft Phone





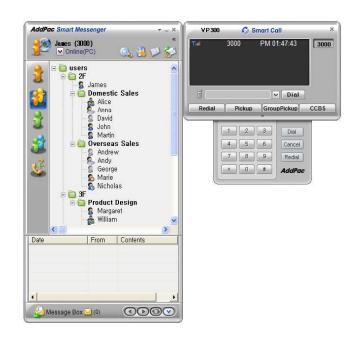
Smart Messenger

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- Support Messenger Service
- Support Various Address Book
- Support User Presence Information
- Support User Search Feature
- Interoperation with Address Book and Smart Phone
- Support Smart Phone Control and Setup
 - Call Control and Forward Setup
- Support Unified Message Box
 - Voice Mail Box
 - Short Message Box





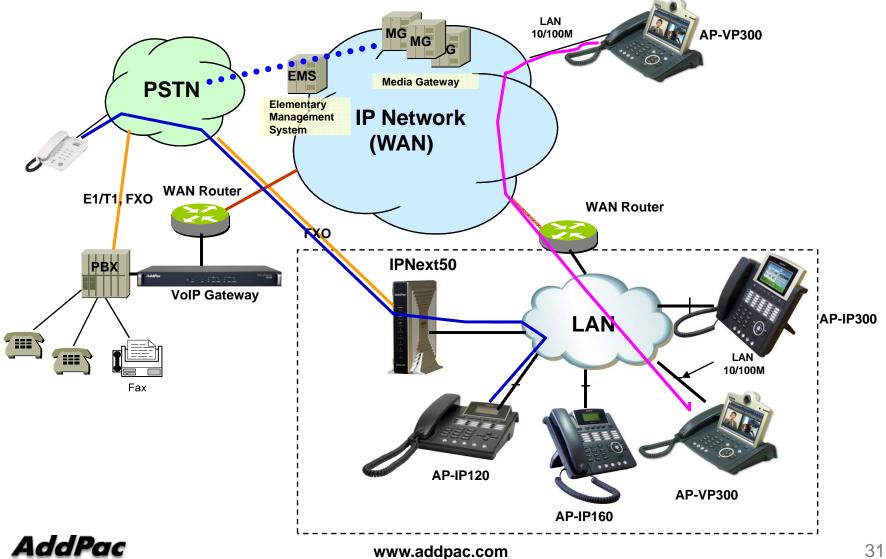


AddPac Smart Messenger

IPNext50 IP-PBX



Standard IP-PBX Application



Ordering Information

• IPNext50 IP-PBX Hardware

- RISC Microprocessor with High-end Programmable DSP Architecture
- 2-ports 10/100Mbps Fast Ethernet(RJ45)
- 1-Port USB Host Mode Interface
- 1-Port RS232C Interface for Command Line Interface
- VoIP Gateway Interface
 Two(2) FXO Ports (IPNext50 Model B)
- Including Network Cable & Ext. Power Supply, etc.
- Built-in APOS Internetworking Software for IPNext50
- Including 1 Year Hardware Warranty
- Product Documents
 - Install and Operation Guide (PDF)
- Pricing
 - AddPac Technology Regional Sales Manager
 - Authorized Sales and Marketing Representatives
 - Please Contact www.addpac.com



IPNext IP-PBX Series

Thank you!

AddPac Technology Co., Ltd. Sales and Marketing

Phone +82.2.568.3848 (KOREA) FAX +82.2.568.3847 (KOREA) E-mail sales@addpac.com

