

AP190 VoIP Gateway



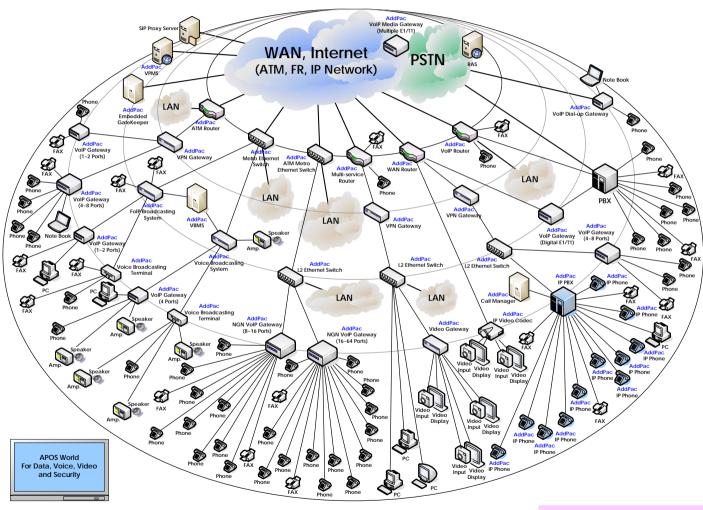
Technical Sales and Marketing

AddPac Technology

www.addpac.com

AddPac VoIP Products Road-Map

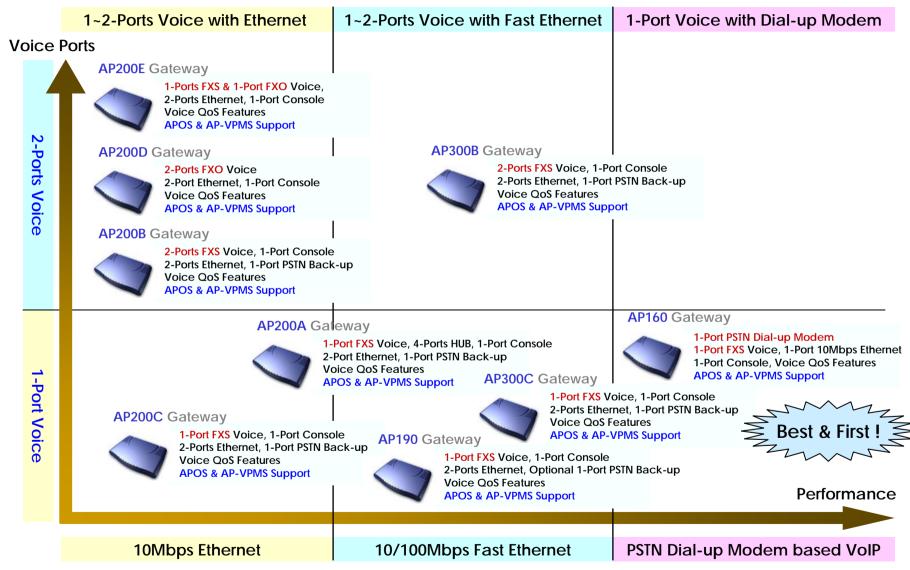
Voice over IP Internetworking Solutions



IP Application Solutions

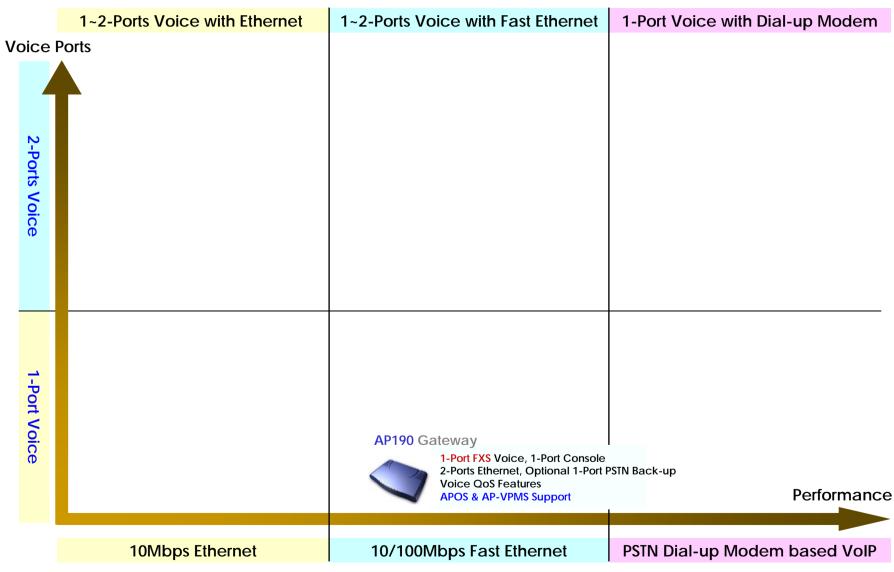


AddPac AP160/190/200/300 Products Road-Map



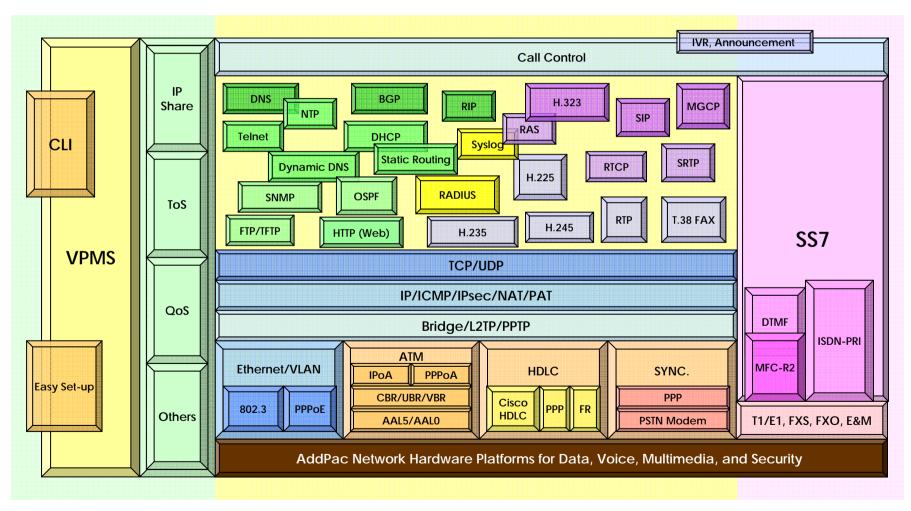


AddPac AP190 Products Road-Map





APOS Internetworking Protocol Stack





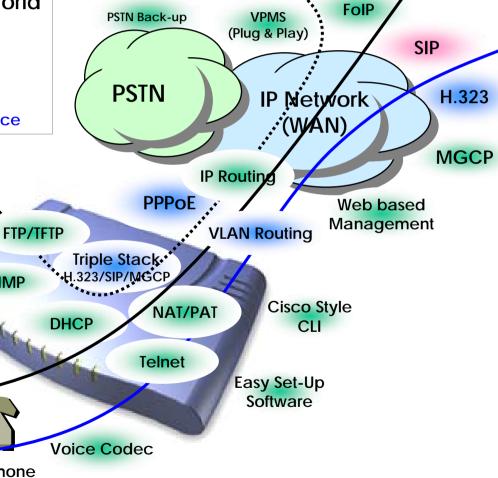
APOS Internetworking S/W World

APOS Internetworking Software World

- AddPac Operating System (APOS)
- **Support Industry Standard**
- Integrated Networking Protocols
- **Optimized Performance & Functionality**
- Easy to use, Installation, and Maintenance

FAX

· IP Sharing





Voice & FAX

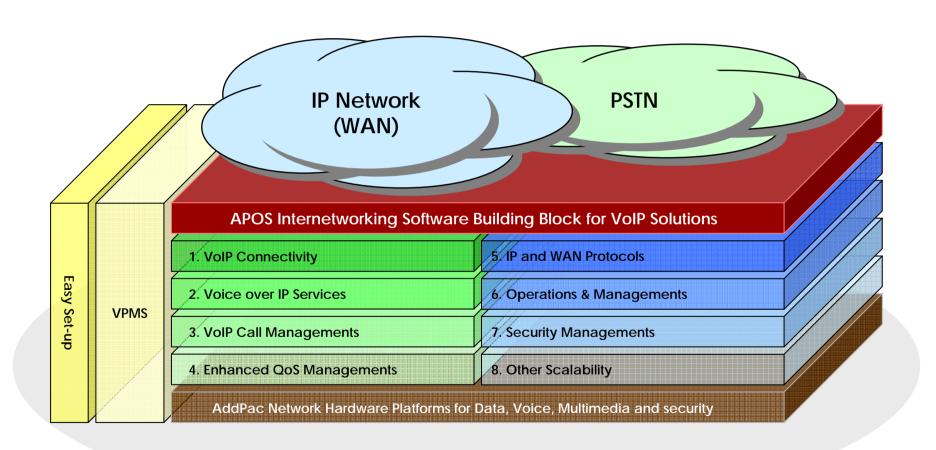
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Data

SNMP

Analog Phone

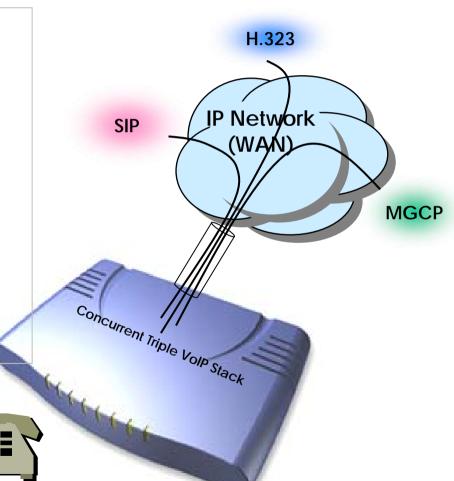
APOS Internetworking S/W World (cont.)





Voice over IP Service

- H.323, SIP, and MGCP Triple Stack
 - Concurrent Triple VoIP Stack Software
 - Single Hardware with Three VoIP Service
 - New APOS-NX Version
- 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
- MGCP
 - IETF RFC2705bis-02 Standard MGCP 1.0





Voice over IP Service (cont.)

• 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
- 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
- DIMETOLO Support with RFC2833 / OPTION
- Re-INVIIE support

MGCP

- Secondary call agent assignment function
- Default package assignment
- Announcement Server Package, Generic Media Package, Handset Package, Line Package, Trunk Package support
- MGCP call agent assignment according to the domain name
- T.38 real-time fax relay support
- DTMF relay support based on RFC2833



VolP

Voice over IP Service (cont.)

Voice Codec

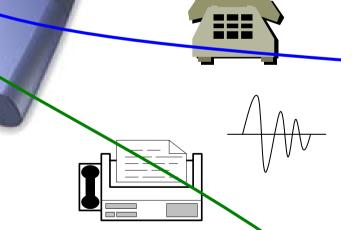
- G.711 A-Law, G.711 U-Law
- G.726 r16, G.726 r32
- G.729A
- G.723.1 r63, G.723.1 r53
- VAD (Voice Activity Detection) function support
- DTMF relay support (H.323, SIP, MGCP common) based on RFC2833

• RTP

- Redundant RTP packet transmission in case of severe packet loss
- Dynamic jitter buffer management and RPI 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

- 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





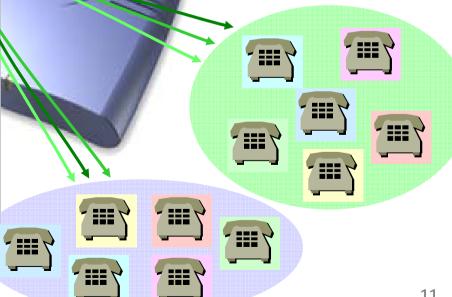
VolP Call Managements

VolP 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

VolP Call Controls (cont.)

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



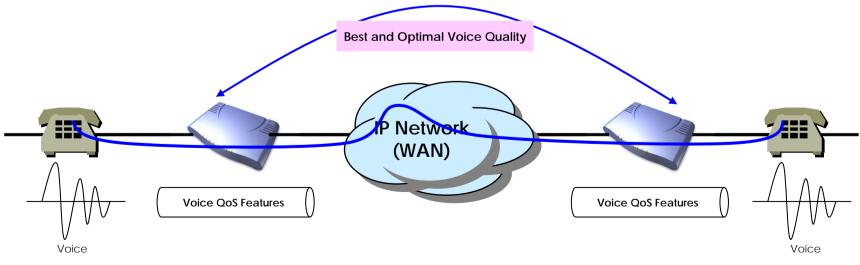


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Enhances QoS Managements

- 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

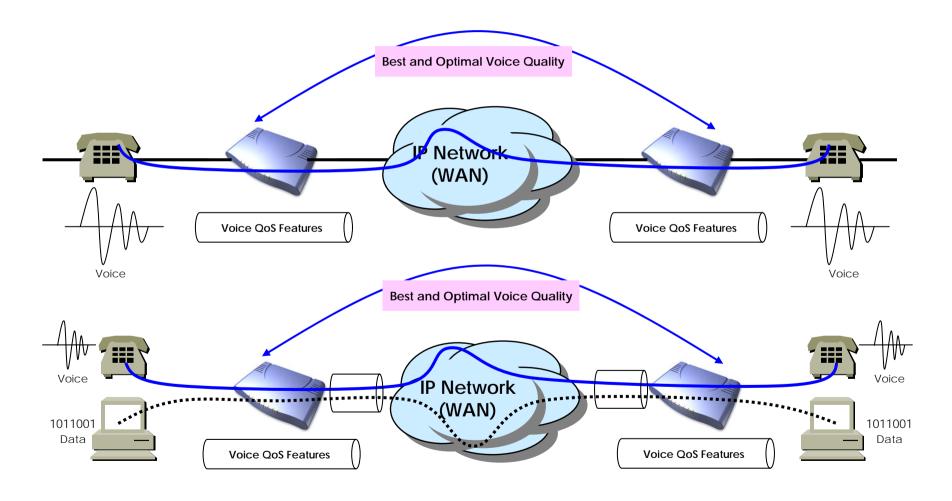
- Enhances Receive Voice QoS Features
 - Dynamic Jitter Buffer Management
 - Error Concealment
 - Support T.38 FAX Data Error Recovery Scheme





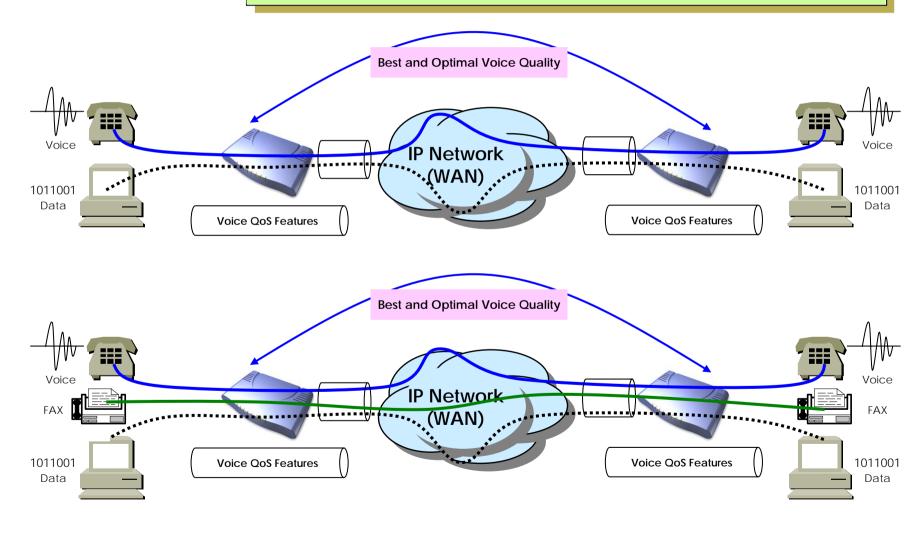
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Enhances QoS Managements (cont.)



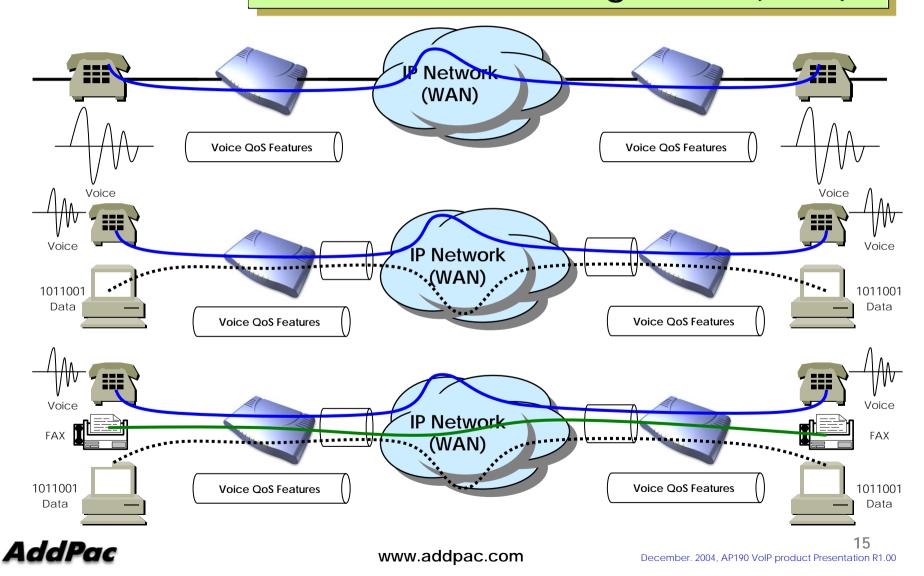


Enhances QoS Managements (cont.)





Enhances QoS Managements (cont.)



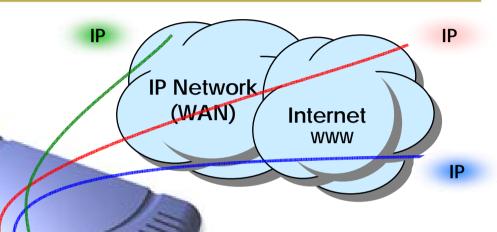
IP and WAN Protocols

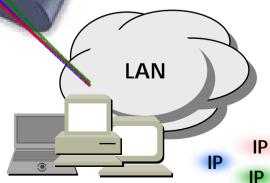
• IP Routing Protocols

- Multi-protocol Internetworking Services
- Static IP routing
- RIP v1/v2 (opt.)
- OSPF v2 routing protocols support (opt)
- Transparent Bridging (IEEE Spanning Tree Protocol)

WAN Protocols

- Point-to-Point Protocol over Ethernet (PPPoE)
- Dynamic Host Configuration Protocol (DHCP) Networking



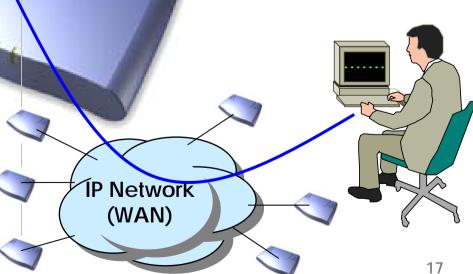




Network Managements

- SNMP
 - Standard Simple Network Management Protocol(SNMP) Agent support
 - MIB v1 and v2 Support
- Web-based Management
 - Standard Voice Interface
 - Standard PSTN Back-up Interface(optional)
- Watch-dog Function
 - Hardware, Software watch-dog services
- Remote Management
 - Telnet
 - Rlogin
 - Console
- Auto Upgrade Service
 - HTTP server based APOS image and configuration file auto-upgrade support
- Batch Job Function
 - Text based script downloading

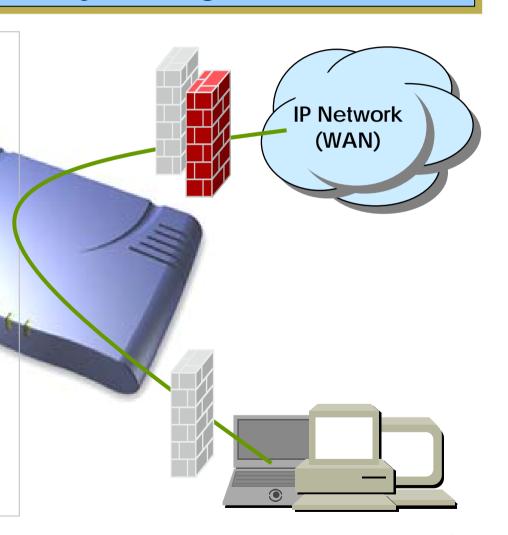
- Easy Set-up Service
 - MS Windows Graphical User Interface (GUI) based installation Software Solution
 - Easy installation Set-up service
- Interoperable with AP-VPMS Service
 - AddPac VoIP Plug & Play Management
 System (AP-VPMS)





Security Managements

- 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 console and 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





Network Scalability

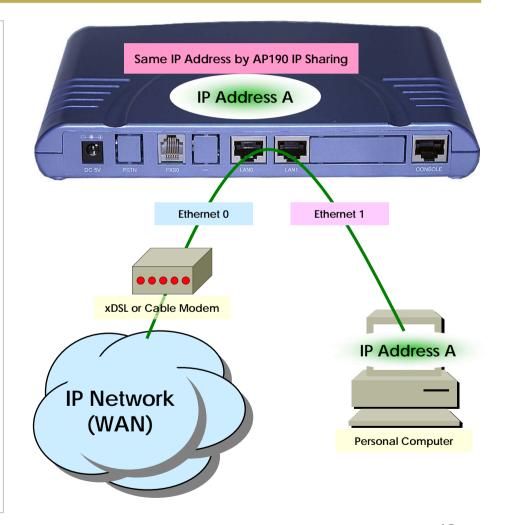
• IP Sharing Feature

- Single IP Address, Multiple IP Network Access
- Public IP Address Sharing by VoIP Gateway

IP Sharing Network Environment

- Support Legacy Network Environment
- xDSL Modem based Broadband Network
- Cable Modem based Broadband Network
- Leased line Network Environment
- Both support Fixed and Dynamic IP
- Support Standard Network Protocols
 - PPPoE
 - PPTP
 - DHCP
- Cascade(Stacking) Function
 - Single Public IP address with stacking multiple VoIP Gateways(virtual ports expansion)

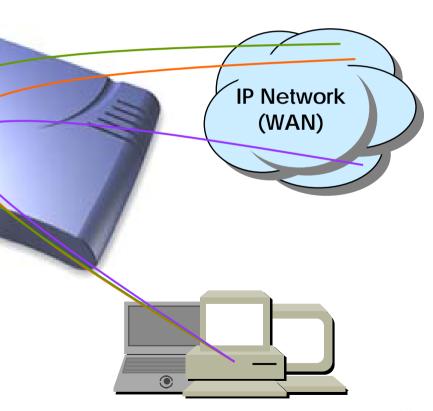
 AP190 VoIP Gateway





Network Scalability (cont.)

- Bridging Service
 - Spanning Tree Bridging protocol (IEEE) support
- DHCP
 - Server and Relay
- IP Accounting
- PAT (Port Address Translation)
- NAT (Network Address Translation)
- Cisco Style CLI (Command Line Interface)
- PPTP (Point-to-Point Tunneling Protocol)
- NTP (Network Time Protocol)
- FTP/TFTP
 - Server and Client support
- DNS
 - Dynamic DNS Client support





VolP Plug & Play System (VPMS)

AddPac AP-VPMS Support

- AddPac VoIP Plug & Play System (VPMS)
- Best EMS Solutions for AddPac VolP 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 Network

(WAN)



AP190 Network Configurations



• Front View Side of AP190



- Rear View Side of AP190
 - 1-Port FXS, 1-Port Console
 - 2-Ports Ethernet,
 - 1-Port PSTN Back-up(Optional)
- AddPac AP-VPMS Support
 - AddPac VoIP Plug & Play System (VPMS)
 - Best EMS Solutions for AddPac VoIP G/W

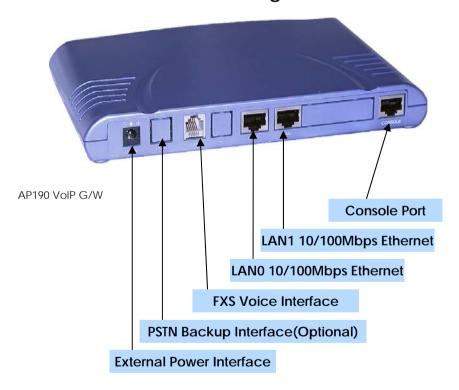


AP190 Hardware Specification

Hardware Specifications

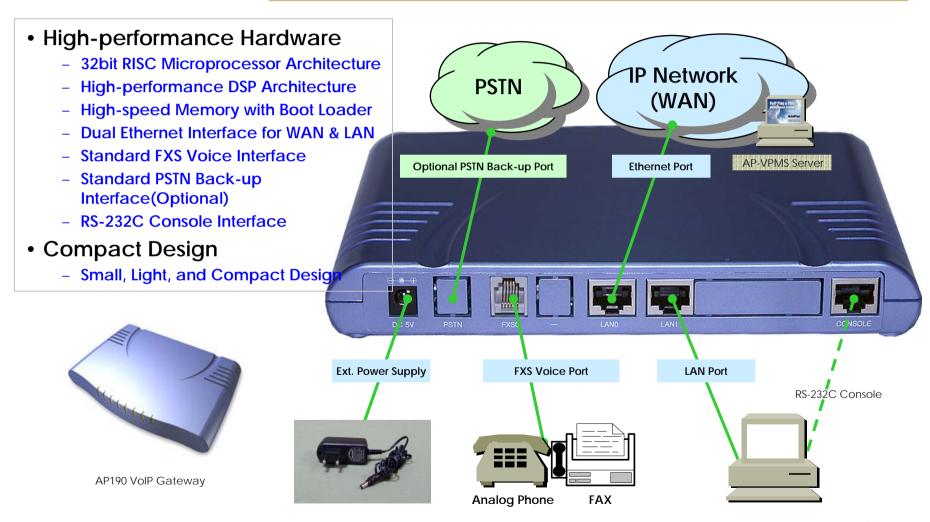
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AP190 VoIP Series	Basic Specifications
CPU	32Bit RISC Microprocessor
Voice Interface	1-Ports FXS Voice Interface(RJ-11)
Ethernet Interface	2-Ports 10/100Mbps Ethernet Interface(RJ-45)
Console Port	1-Port RS-232C Console Port(RJ-45)
PSTN Backup Port(Optional)	1-Port PSTN Backup Port(RJ-11)
Flash Memory	2Mbyte High-speed Flash Memory
Base Memory	16Mbyte High-speed SDRAM
Power Requirement	External Power Supply Adaptor / VAC 110~220V, 50/60Hz, 10Watt
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	26mm x 200mm x 137mm (H x W x D)
Weight (g)	400g

Network interface Configurations



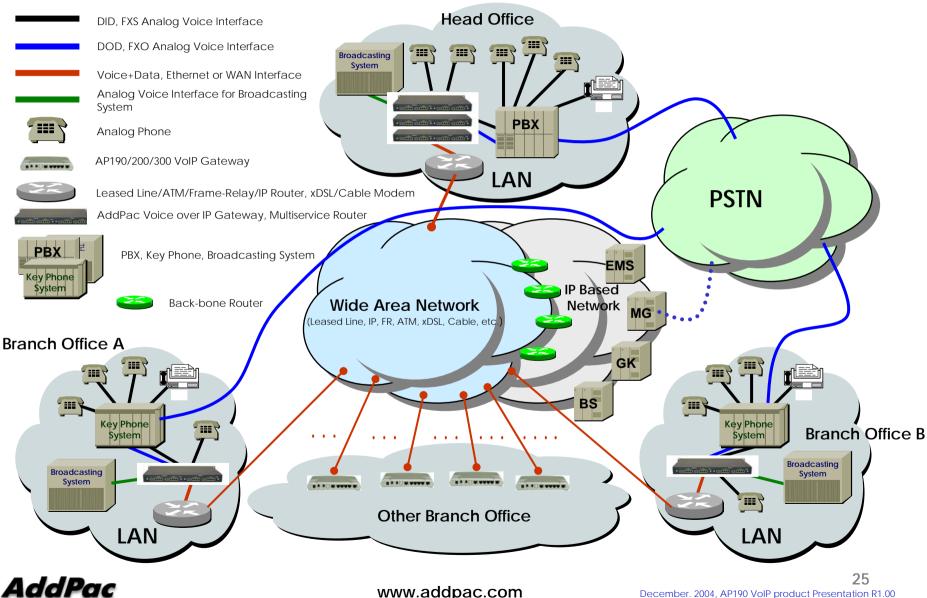


AP190 Networking Interface

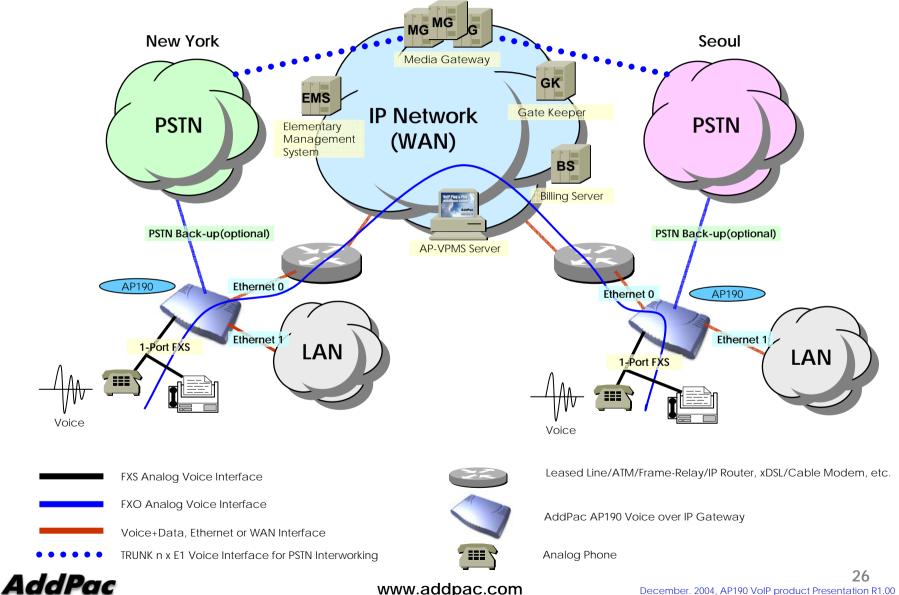




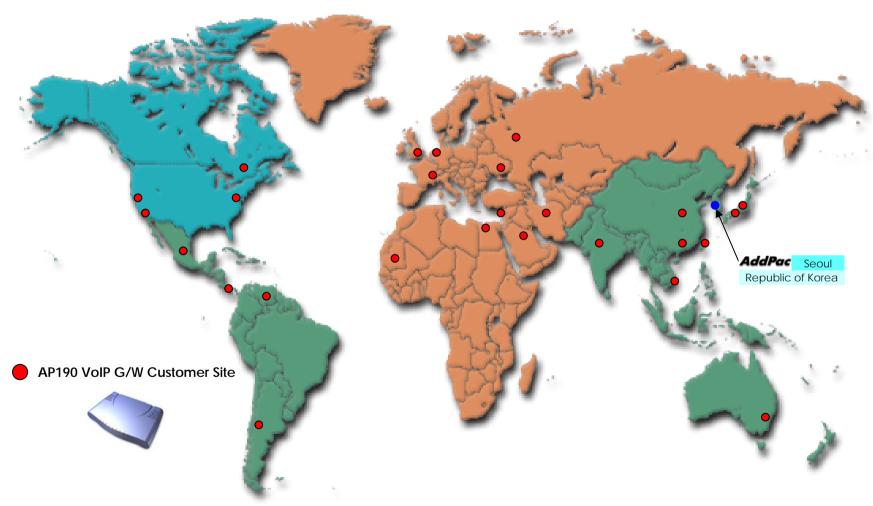
AddPac VoIP Gateway Network Diagram Example



AP190 Voice Networking Configuration



AP190 World-Wide Customer Site





AP190 Ordering and Pricing

AP190 VoIP Gateway Series

- VolP Gateway Hardware
 - High-speed 32bit RISC CPU
 - 1-Port FXS Voice port Configuration
 - 2-Ports 10/100BaseT Ethernet
 - Optional 1-Port PSTN Back-up
 - 1-Port RS-232C Console
 - Including Network Cable Set
- Built-in APOS Internetworking Software
- Including 1-Year Hardware Warranty
- Product Documents
 - Installation Guide (PDF)
 - Quick APOS Operation Guide (PDF)
 - APOS Operation Guide (PDF)

• Pricing

- AddPac Technology Regional Sales Managers
- Authorized Sales and Marketing Representatives
- Please Contact www.addpac.com



OEM and volume pricing available!



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