



AP-GS716™

16-Port GSM VoIP Gateway

High Performance GSM VoIP Gateway Solution

Preliminary Product Overview

(Without notice, following described technical spec. can be changed)



AddPac

AddPac Technology

2012, Sales and Marketing

www.addpac.com

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- Hardware Specification
- APOS Technology
- VoIP (Voice over IP) service
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- Network Protocols
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Product Overview

AP-GS716 16-Port GSM VoIP Gateway

- 16-Port GSM VoIP Gateway Solution
- 16-Port SIM Slots
- 16-Port Antenna Interface
- GSM VoIP Interface(LAN) Support
- H.323/SIP Dual Concurrent Stack Embedded
- High Performance RISC & Programmable DSP Architecture
- G.711/G.726/G.723/G.729, T.38 Fax , VAD, etc
- Powerful Network Protocols (PPPoE, DHCP, Static Routing, etc)
- Two(2) 10/100Mbps Fast Ethernet (IP Share ,etc)
- One(1) RS-232C Port for Command Line Interface
- Firmware Upgradeable Architecture
- Smart Web based Manager Software
- Advanced Voice QoS Mechanism
- Light and Compact Design with Internal Power Supply

Product Highlights

AP-GS716 16-Port GSM VoIP Gateway

High Performance GSM VoIP Gateway

State-of-art Signaling
H.323, SIP
Concurrent Dual Stack

Up to 16-Port GSM Interface

Advanced Smart Web
Manager Support

High Performance RISC CPU
+ Programmable DSP

Excellent Voice Quality
G.729/G.723/G726/G711

Broadband IP Networking
With two(2) 10/100Mbps
Fast Ethernet

APOS™ Technology
Firmware Upgradeable Architecture

Smart NMS(Network Management
System) Support

Powerful Network Protocol
Support

Powerful Trouble Shooting &
Debugging Feature Support

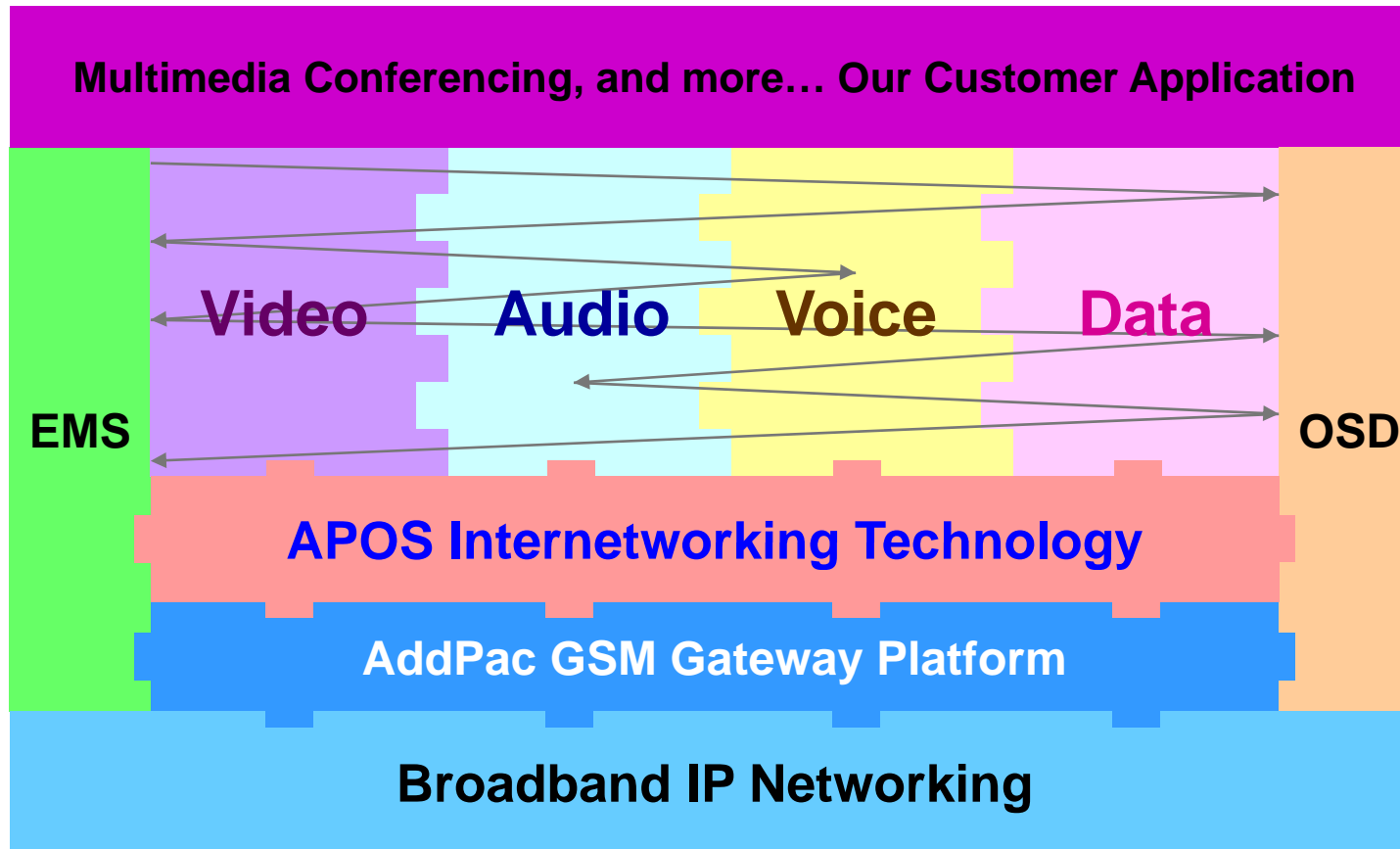
Advanced Voice
Traffic QoS Mechanism

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APOS Technology

AP-GS716 16-Port GSM VoIP Gateway



- APOS : AddPac Internetworking Operating System
- OSD : On- Screen Display
- EMS : Element Management System

Hardware Specification

AP-GS716 16-Port GSM VoIP Gateway

RISC
CPU

High-end
DSP

- RISC Microprocessor Computing Power
- Powerful High-End DSP for VoIP Interface
- 16-Port GSM VoIP Gateway
- 16-Port SIM Card Slots
- 16-Port GSM Antenna Interface
- Network Interface for VoIP Direct Interface
 - Two(2) 10/100Mbps Fast Ethernet (RJ45)
- RS232C Console Interface for CLI
- Run LED, LAN LED, Port LEDs at Front Side
- Compact and Light Design with Internal Power Supply
- Small and Short Antenna

Hardware Specification

AP-GS716 16-Port GSM VoIP Gateway

RISC
CPU

High-end
DSP

| AP-GS716 GSM VoIP Gateway | Basic Specifications |
|---------------------------|--|
| GSM VoIP Interface | 16-Port GSM VoIP Interface |
| GSM Antenna Interface | 16-Port GSM Antenna Interface |
| SIM Card Slot Interface | 16-Port GSM SIM Card Slots |
| Ethernet Interface | Two(2) 10/100Mbps Fast Ethernet (RJ-45) |
| Console Interface | One(1) RS-232C Interface for CLI (RJ45) |
| Flash Memory | 512Mbyte NAND Flash |
| Main Memory | 128Mbyte DDR2 Memory |
| Power Requirement | Power Supply Adaptor / VAC 110~220V, 50/60Hz, 5V 20A |
| 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) |
| Dimension (H x W x D) | (60mm x 340mm x 268mm) - 19" Rack Mountable Chassis |
| Weight | 3.2Kg |

Hardware Specification

AP-GS716 16-Port GSM VoIP Gateway

RISC
CPU

High-end
DSP

Front Side View



LAN0 10/100Mbps
Ethernet

LAN1 10/100Mbps
Ethernet

RS232C
Console Port

Port LEDs

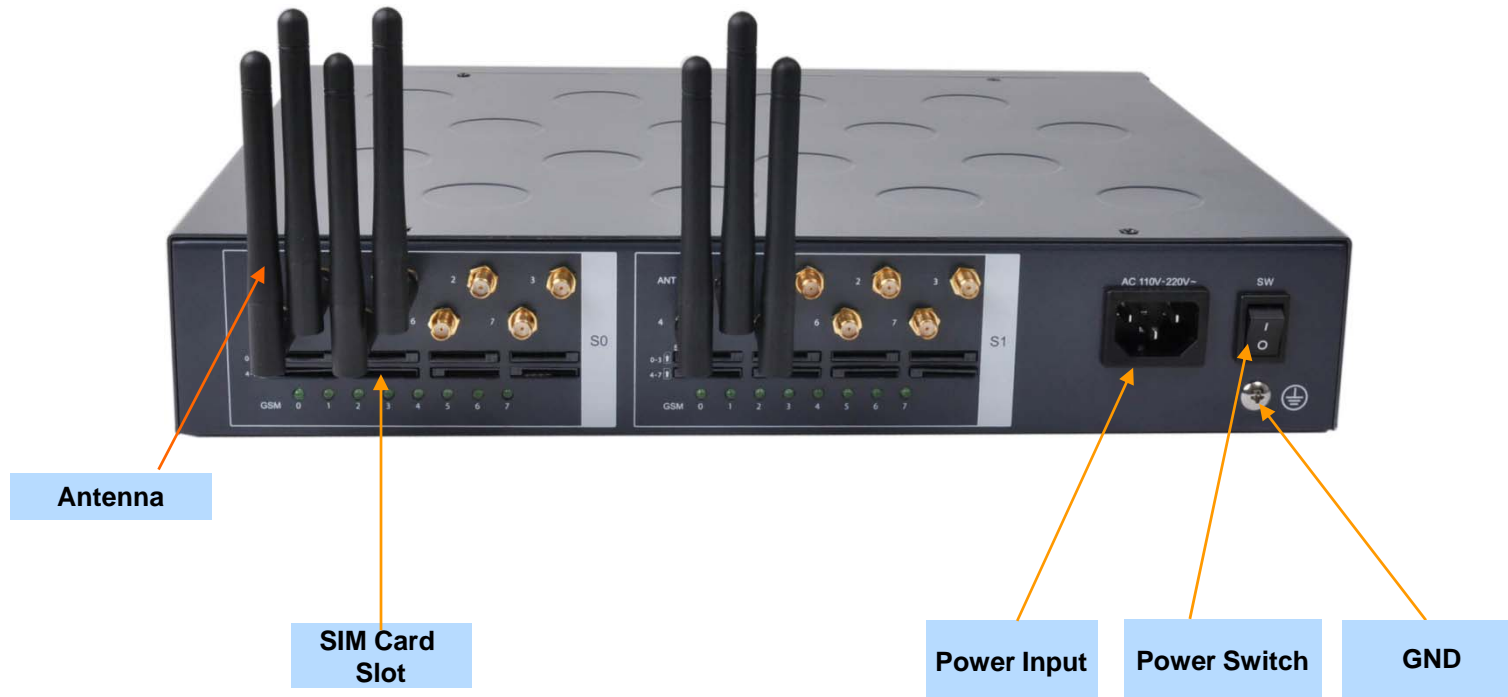
Hardware Specification

AP-GS716 16-Port GSM VoIP Gateway

RISC
CPU

High-end
DSP

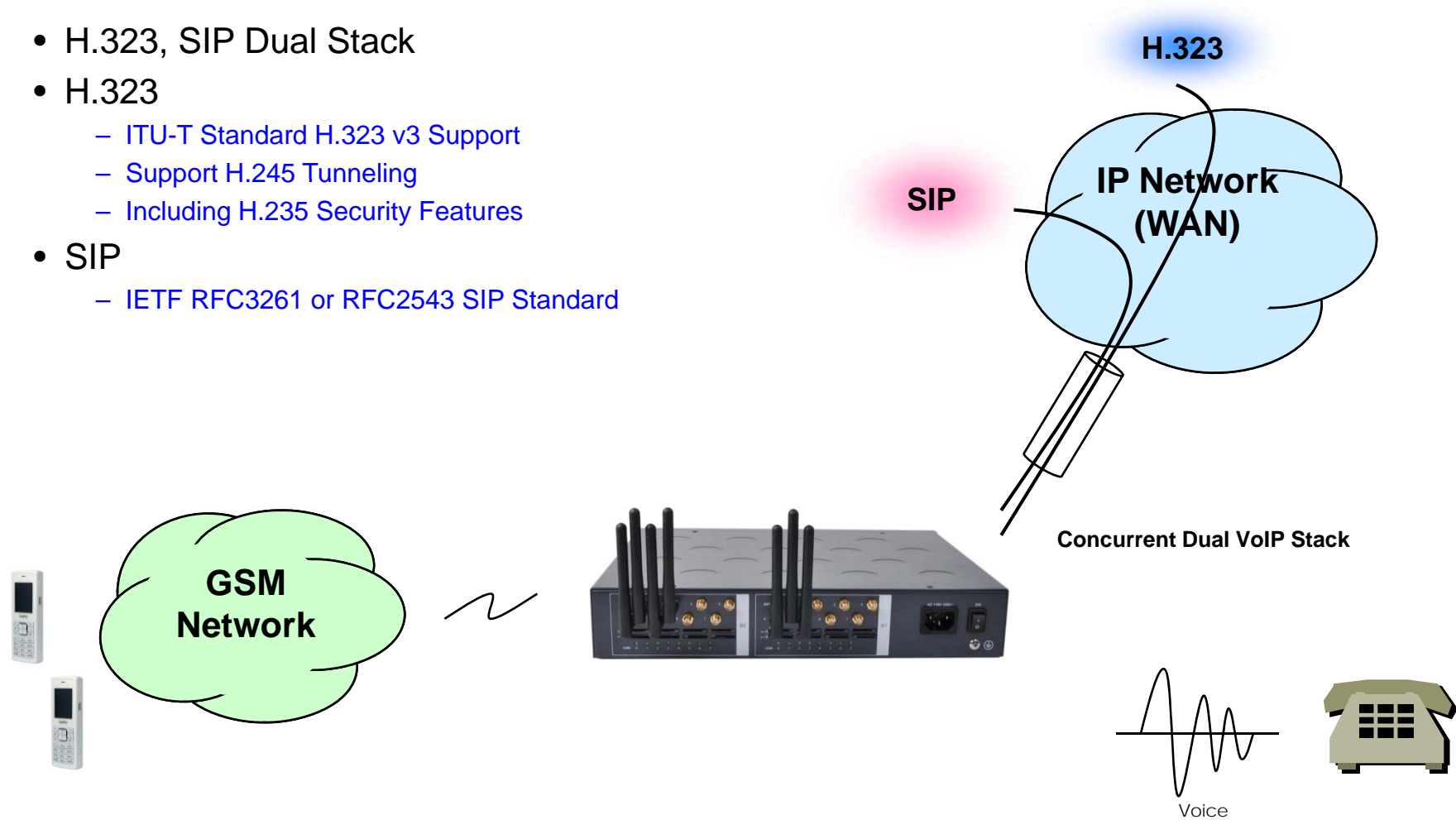
Back Side View



VoIP (Voice over IP) Service

AP-GS716 16-Port GSM VoIP Gateway

- H.323, SIP Dual 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



VoIP (Voice over IP) Service

AP-GS716 16-Port GSM VoIP Gateway

- 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

VoIP (Voice over IP) Service

AP-GS716 16-Port GSM VoIP Gateway

- 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

VoIP



GSM Network

VoIP (Voice over IP) Service

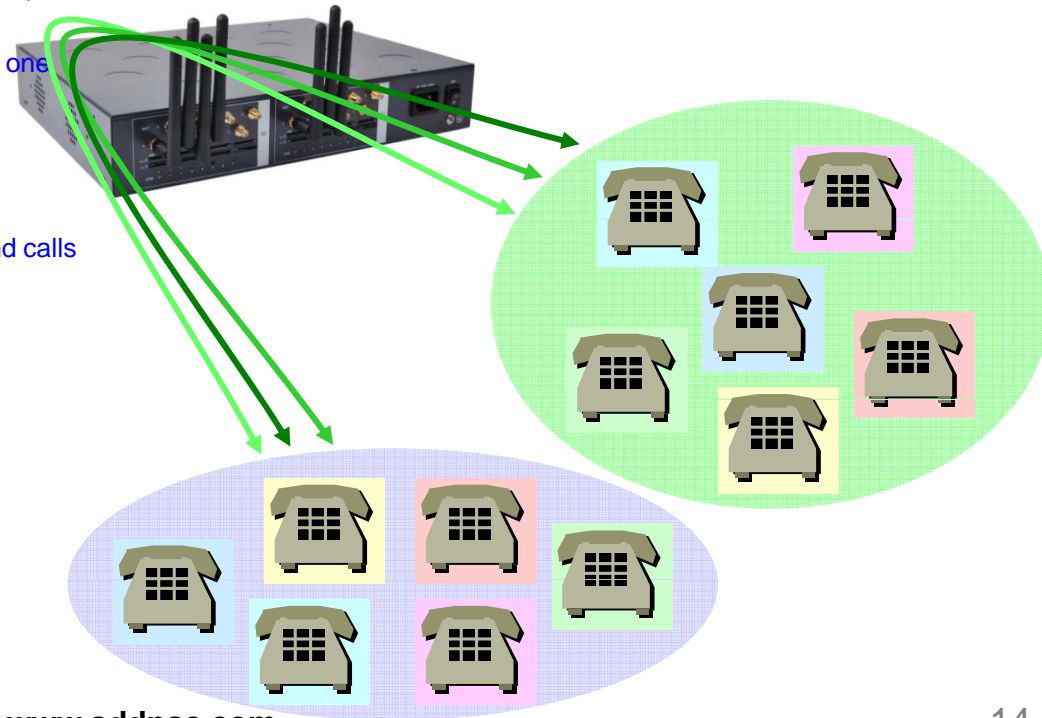
AP-GS716 16-Port GSM VoIP Gateway

- 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



Advanced QoS Features

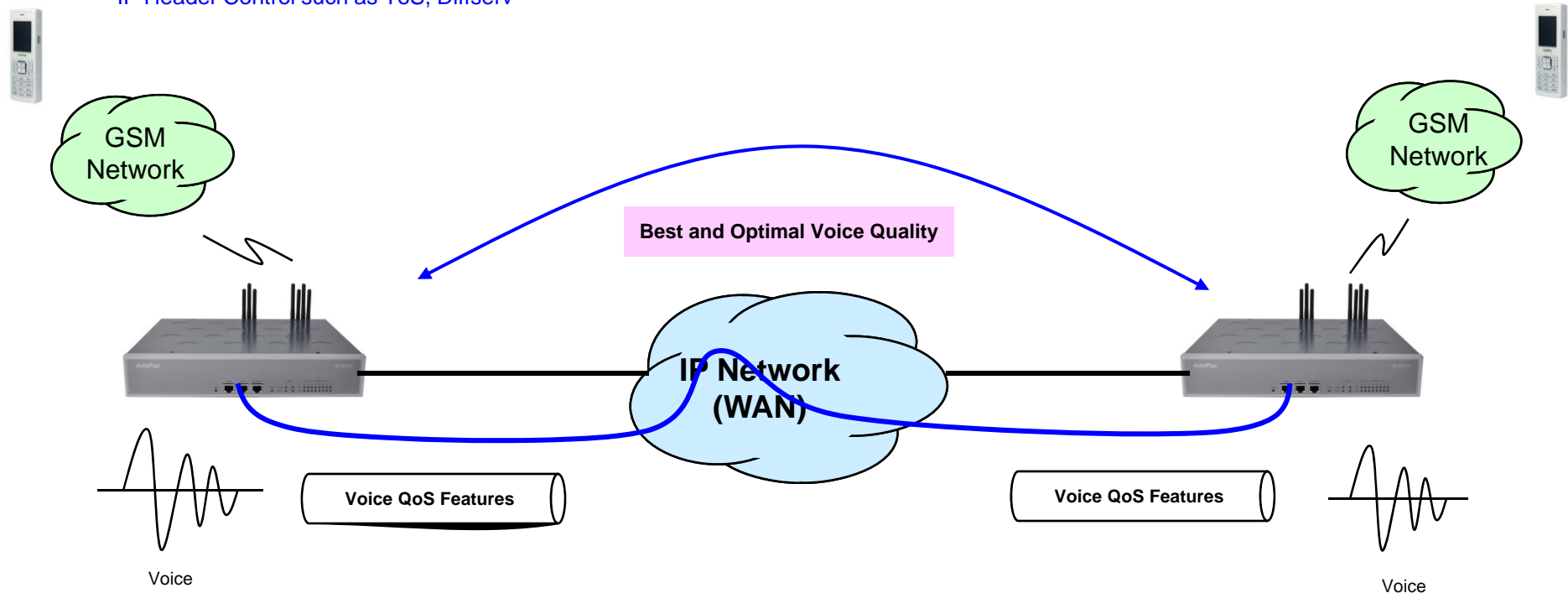
AP-GS716 16-Port GSM VoIP Gateway

- 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

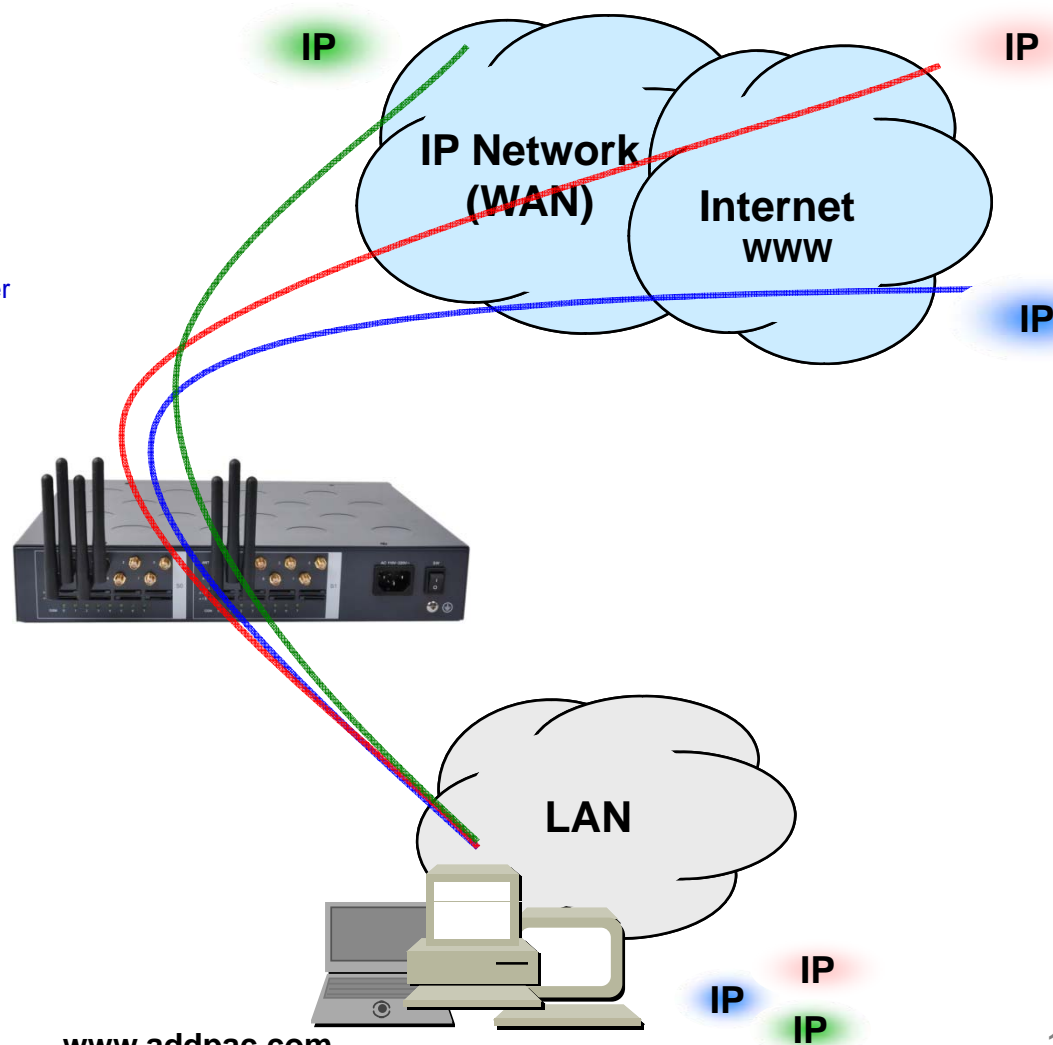
- Dynamic Jitter Buffer Management
- Error Concealment
- Support T.38 FAX Data Error Recovery Scheme



Network Protocols

AP-GS716 16-Port GSM VoIP Gateway

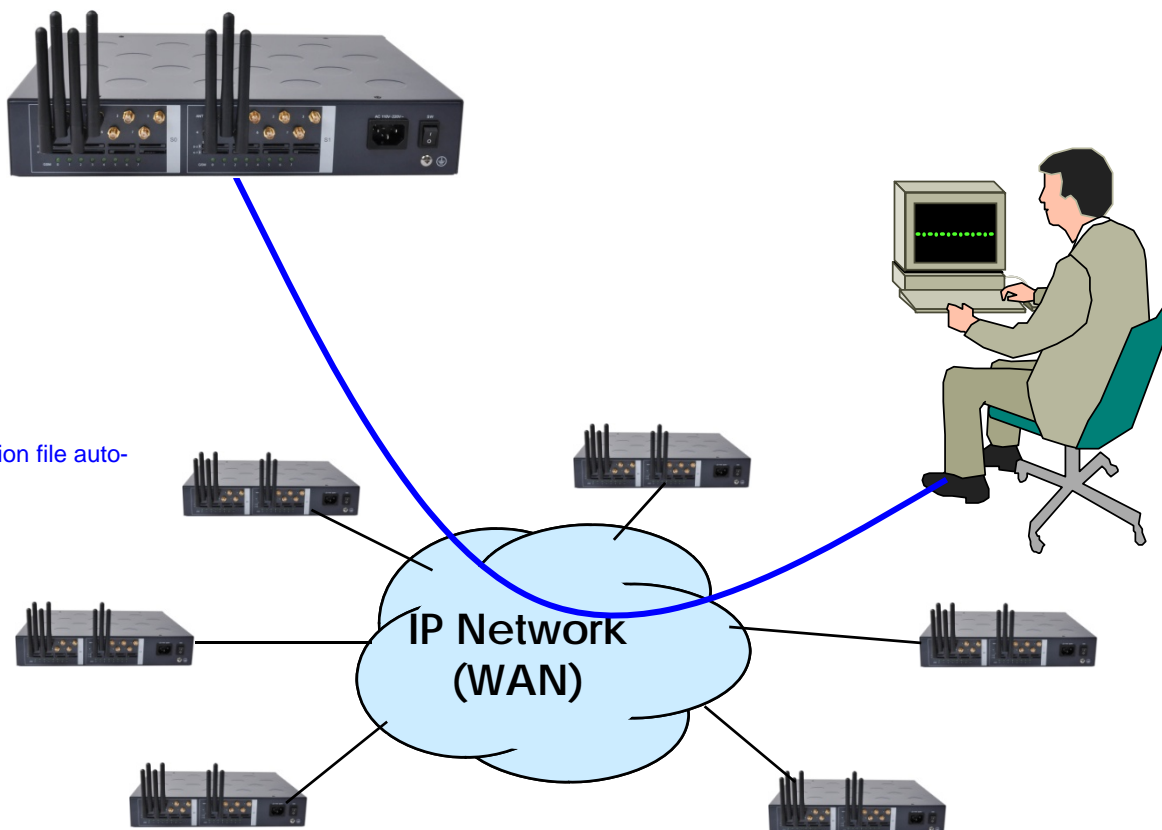
- **Basic Network Protocols**
 - ARP, IPv4, TCP, UDP, ICMP, SCTP, IGMP, MLD
- **Routing Protocol**
 - IPv4 : Static
- **Service Protocol**
 - FTP, Telnet, TFTP, DHCP Server/Relay, SNMP Server
 - CDP (Cisco Discovery Protocol)
 - DNS Resolver, DDNS (nsupdate)
 - Bridge
 - Syslog
- **IPv4 Address Configuration**
 - Fixed (Static)
 - DHCP
 - PPPoE
- **Miscellaneous**
 - Cisco Style CLI
 - Standard & Extended IPv4 Access List
 - Multi-level User Account Management
 - IP accounting
 - STUN Client



Network Management

AP-GS716 16-Port GSM VoIP Gateway

- **SNMP**
 - Standard Simple Network Management Protocol(SNMP) Agent support
 - MIB v1 and v2 Support
- **Web-based Management**
 - Smart Easy Setup
 - Standard Voice Interface
 - Standard PSTN Back-up Interface
- **Watch-dog Function**
 - Hardware, Software watch-dog services
- **Remote Management**
 - Telnet
 - Rlogin
- **Auto Upgrade Service**
 - HTTP server based APOS image and configuration file auto-upgrade support
- **Batch Job Function**
 - Text based script downloading
- **Interoperable with AP-VPMS Service**
 - AddPac VoIP Plug & Play Management System (AP-VPMS)



Smart Web Manager : Main Page Layout

AP-GS716 16-Port GSM VoIP Gateway

Main Menu
For easy system setup, provide the various menu and category

- System
 - Network Setup
 - Language
 - NTP
 - Backup/Restore
- Basic
 - Protocol
 - Server SIP
 - Server H.323
 - Tel Number
 - FXS/FXO/E1 Group
 - E1 Trunk
 - DTMF/CODEC
 - Dial Plan/Prefix
 - Static Route
 - Hot Line
- Advanced
 - Gain
 - Fax
 - Service
 - Filtering
 - Security
 - SNMP
- Miscellaneous
 - Call Status
 - System Status
 - Alarm Status
 - Call Log
 - System Log
 - Test Call
 - Ping

Tool Bar
Provide frequently used tools like as System Update, Configuration Backup, Initialization, Restart, Telnet

System Information

| | |
|------------------------|--|
| HW Version | 2.0 |
| S/W Version | ap1800k_web_g2_v8_47T.bin 8.47 |
| MAC Address | 0002.a511.2245 |
| VoIP Protocol | SIP |
| Voice Interface Module | S(4)O(4) : E1(2) |
| Registration Status | Unregistered |
| Supported Codec List | g711alaw g711ulaw g7231r53 g7231r63 g726r32 g729 |
| Network Information | Static 172.16.50.114 |
| WAN LINK Status | 100Mbps FULL Duplex Link UP |
| LAN LINK Status | Link Down |
| Current Time | Thu Oct 1 13:06:23 2009 |
| System Startup Time | Thu Oct 1 12:56:46 2009 |
| System Running Time | 0 days 00:09:37 |
| Total Calls | 0 |

Information
Display the current system version and status summary

Information

AddPac Tehonology
Model : AP1800K_G2
HW Version : 2.0
S/W Version : 8.47
Smart Web Version : 0.3
Smart Web Build : Oct 1 2009
Voice Interface
S(4)O(4) : E1(2)
Protocol : SIP
Status : Unregistered
CurrentCalls : 0 Call
Network : Static 172.16.50.114
Mac Address: 0002.a511.2245

Description

WAN 포트에 대한 설정입니다. Static IP의 경우 고정 IP 주소로 사업자로부터 할당 받은 주소 정보를 입력합니다. DHCP와 PPPoE의 경우 유동 IP로 장비의 주소가 변경될 수 있습니다. DHCP 및 PPPoE는 사용자 환경에 맞도록 설정하십시오. MAC 주소 변경은 필요시 장비에 설정된 주소를 사용하지 않고 사용자가 설정한 주소를 사용하는 방안으로 반드시 필요한 경우에 한하여 사용하여야 합니다.

Workspace
Workspace for detailed action

Description
Display the help message if you move mouse over main menu

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Smart Web Manager : SIP Server (Example)

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Smart Web Manager
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SIP (Session Initiation Protocol)

Use SIP Server Yes No

Primary SIP Server Server address (IP or Domain Name) and Port (default 5060)

Secondary SIP Server Server address (IP or Domain Name) and Port (default 5060)

Local Domain name (SIP userpart of authentication)

SIP Signaling Port (default 5060, between 1 to 65535)

Register Expiration (in seconds, default 60, between 10 to 86400)

Session Re-Fresh INVITE UPDATE

Session Expire Time (in seconds, default 1800, between 30 to 86400, 0 = disable)

Apply

SIP Server
Primary & Secondary server,
Local domain name,
SIP Signaling Port (**reboot necessary**)
Timer
* register expire
* session refresh
* session expire

Information
AddPac Technology
Model : GS1002_G2
H/W Version : 2.0
S/W Version : 8.00d
Smart Web Version : 0.4
Smart Web Build : Mar 24 2010
Voice Interface
G(2)S(2)
Protocol : SIP
Status : Unregistered
CurrentCalls : 0 Call
Network : Static 172.16.9.16
Mac Address : 0002.a400.0000
Unread Message:
P0:0(0)
P0:1(0)

Description
Configure the settings for SIP.
Contact your service provider
for the settings

Smart Web Manager : GSM Extension (Example)

AP-GS716 16-Port GSM VoIP Gateway

The screenshot shows the Smart Web Manager interface for configuring GSM extensions. The main content area is titled "GSM Extension" and is divided into three sections: "Port Information", "GSM Extension Configuration", and "GSM Extension with Translation".

Port Information: A table showing port configurations for P0, P1, P2, and P3.

| Port | P0 | P1 | P2 | P3 |
|-------|-----|-----|-----|-----|
| SLOT0 | GSM | GSM | FXS | FXS |

GSM Extension Configuration: A table for configuring individual extensions.

| Index | Port | Numbers | Preference | HuntStop | Select |
|-------|------|---------|------------|----------|--------------------------|
| 0 | 0/0 | T | 0 | X | <input type="checkbox"/> |

GSM Extension with Translation: A table for defining translation rules.

| Port | Destination Pattern | Digits to Insert | Number of Digits to Delete |
|------|---------------------|------------------|----------------------------|
| P0:0 | 33 | 8 | 1 |
| P0:1 | | | 0 |

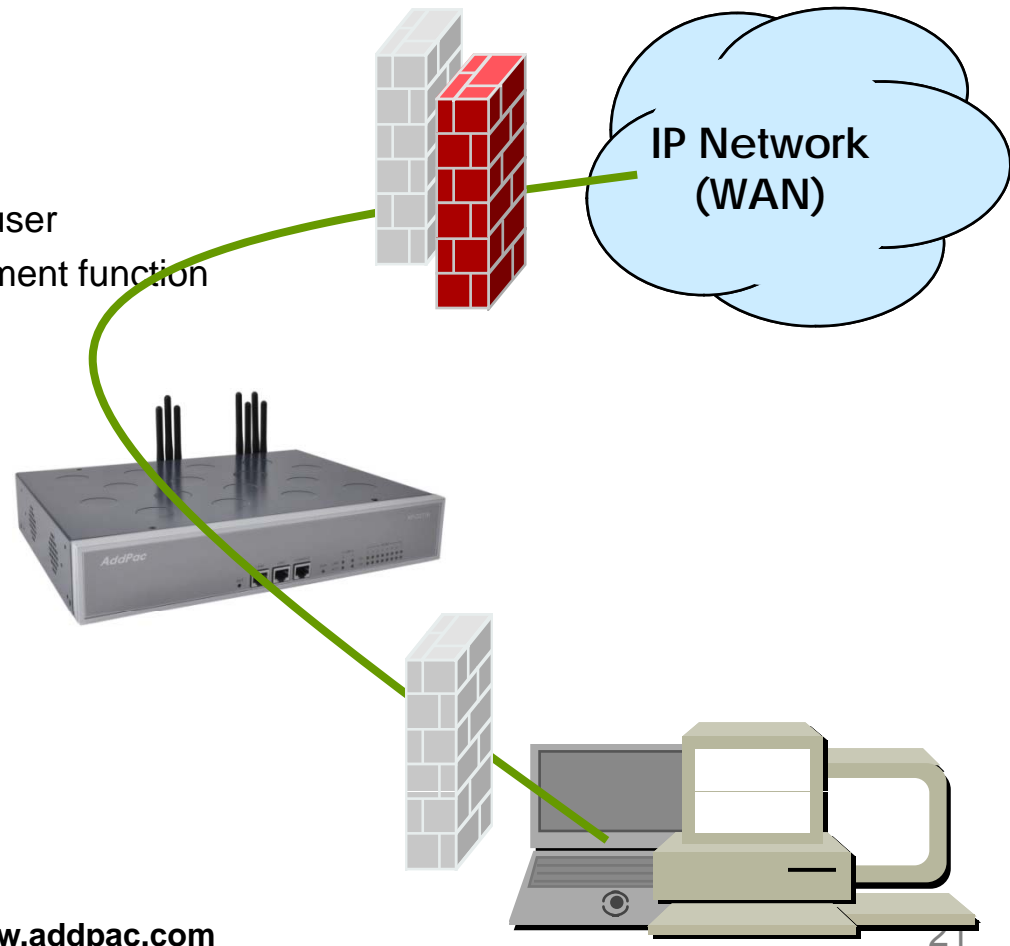
Callouts provide additional context:

- Port Information:** voice port type & physical port
- GSM Extension:** Configure GSM phone-number for receiving a call (usually 'T' is used for each port)
- GSM Extension with Translation:** Used to GSM callback
 - The Received CID is not real serving number.
 - The specified translation rule is applied.

Security Management

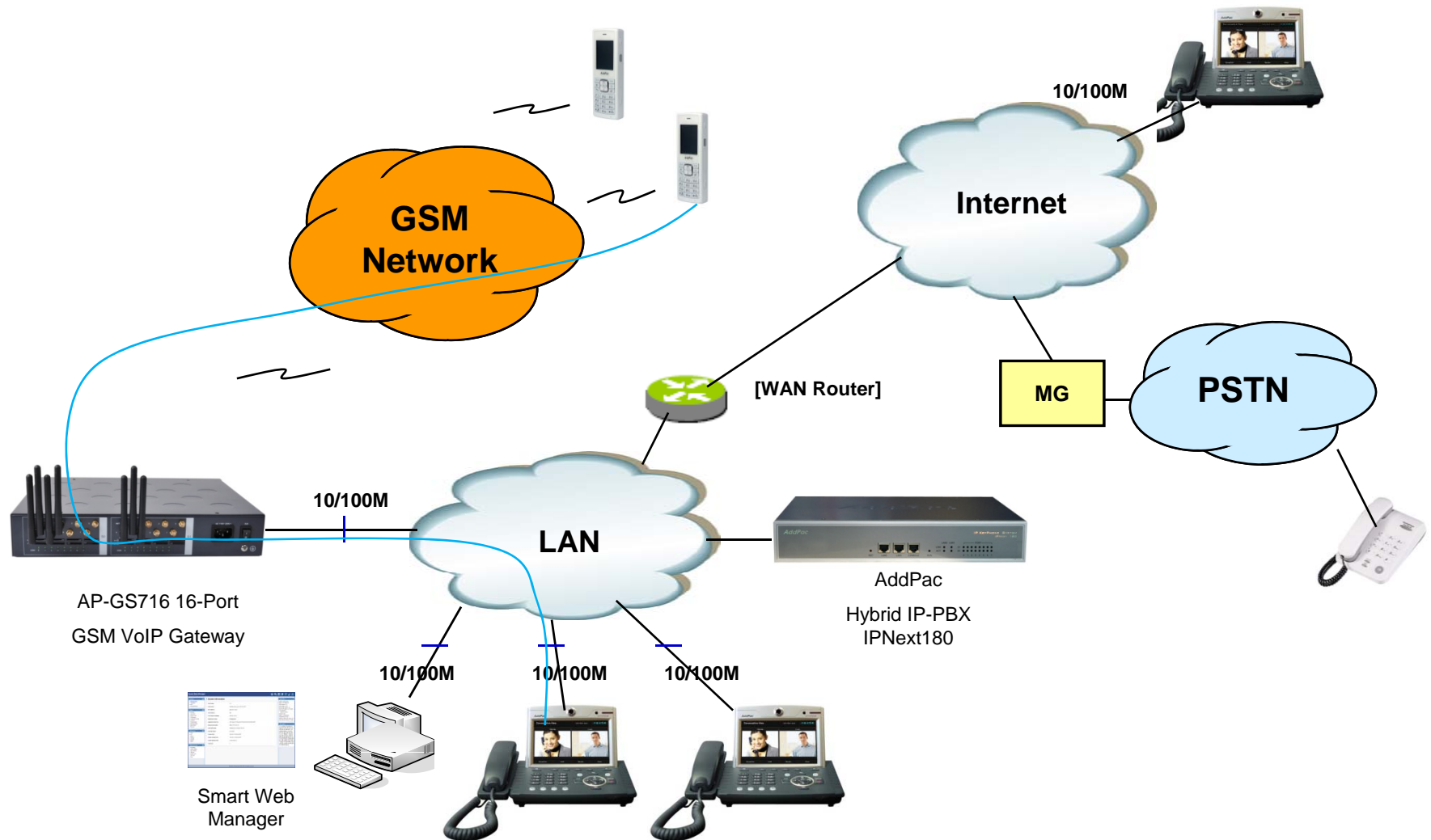
AP-GS716 16-Port GSM VoIP Gateway

- 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



Standard Application

AP-GS716 16-Port GSM VoIP Gateway



Ordering Information

- AP-GS716 GSM Gateway Hardware
 - AP-GS716 GSM VoIP Gateway Main Body
 - RISC Microprocessor with High-end Programmable DSP Architecture
 - 2-port 10/100Mbps Fast Ethernet(RJ45)
 - 1-Port RS232C Console Interface(RJ45)
 - Including Network Cable, Antenna, etc.
- Built-in APOS Internetworking Software for AP-GS716
- 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

GSM VoIP Gateway 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