



# AP-GS701E

## 1-Port GSM Gateway

(1-Port GSM, 1-Port FXS, 1-Port FXO)  
High Performance GSM Gateway Solution



### Preliminary Product Overview

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

**AddPac**

**AddPac Technology**

Sales and Marketing

[www.addpac.com](http://www.addpac.com)

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# Product Overview

## AP-GS701E 1-Port GSM Gateway

- Analog Interface (FXS, FXO)/VoIP Interface(LAN) Both Support
- 1-Port GSM, 1-Port FXS, 1-Port FXO Interface
- H.323/SIP Dual Concurrent Stack Embedded
- High Performance RISC & Programmable DSP Architecture
- One(1) 10/100Mbps Fast Ethernet (IP Share ,etc)
- High Performance LAN-to-LAN Routing Capability
- G.711/G.726/G.723/G.729, T.38 Fax , VAD, etc
- Powerful Network Protocols (PPPoE, DHCP, Static Routing, etc)
- Firmware Upgradeable Architecture
- VPMS (VoIP Plug&Play Management System) for Large Scale Deployment
- Advanced Voice QoS Mechanism
- Light and Compact Design with External Power Supply

# Product Highlights

## AP-GS701E 1-Port GSM Gateway

### Powerful GSM Gateway

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

1-Port GSM Interface  
1-Port FXO Interface  
1-Port FXS Interface

Analog interface,  
VoIP Interface Both  
Support

High Performance RISC CPU  
+ Programmable DSP

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

Broadband IP Networking  
With one(1) 10/100Mbps  
Fast Ethernet

APOS™ Technology  
Firmware Upgradeable Architecture

VPMS Large Scale VoIP  
Gateway Management Support

Powerful Network Protocol  
Support

Powerful Trouble Shooting &  
Debugging Feature Support

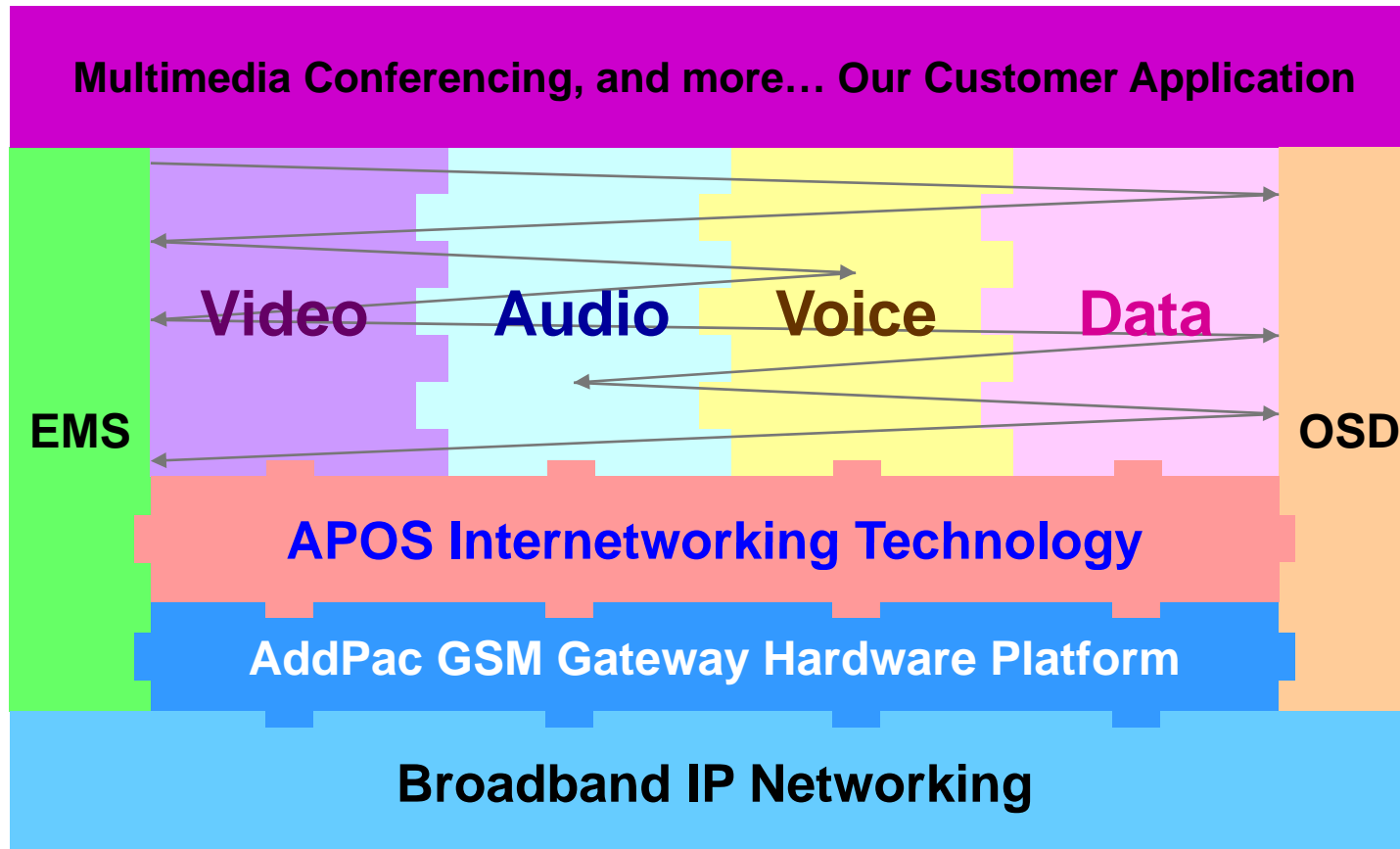
Advanced Voice  
Traffic QoS Mechanism

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# APOS Technology

AP-GS701E 1-Port GSM Gateway



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

# Hardware Specification

AP-GS701E 1-Port GSM Gateway

RISC  
CPU

High-end  
DSP

- **RISC Microprocessor Computing Power**
- **1-Port GSM Gateway**
- **1-Port SIM Card Slot**
- **1-Port GSM Antenna Interface**
- **VoIP Gateway Interface**
  - One(1) FXS Port
  - One(1) FXO Port
- **Network Interface for VoIP Direct Interface**
  - One(1) 10/100Mbps Fast Ethernet (RJ45)
- **Run LED, LAN LED, Port LEDs**
- **External Power Supply**

# Hardware Specification

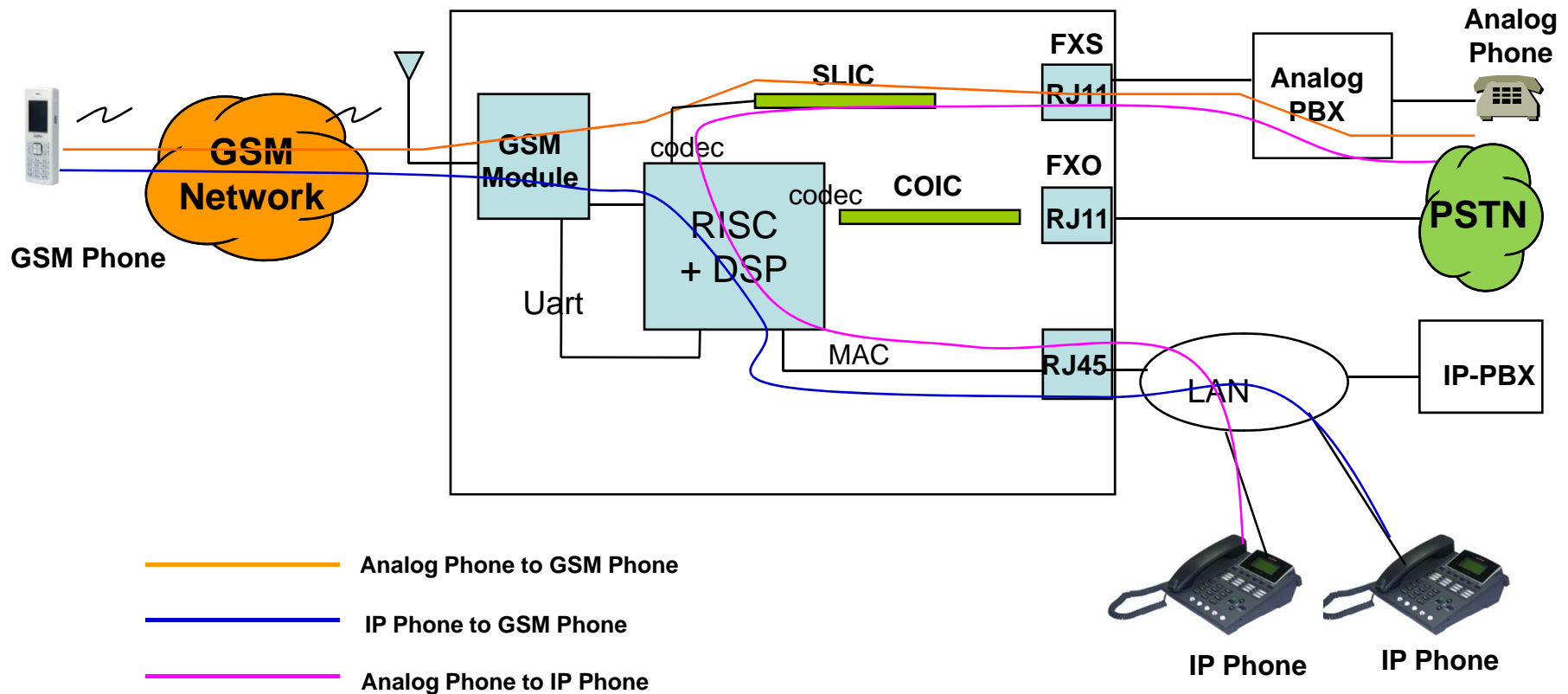
AP-GS701E 1-Port GSM Gateway

RISC  
CPU

High-end  
DSP

## GSM Gateway Service Diagram

AP-GS701E GSM Gateway Internal H/W Block Diagram



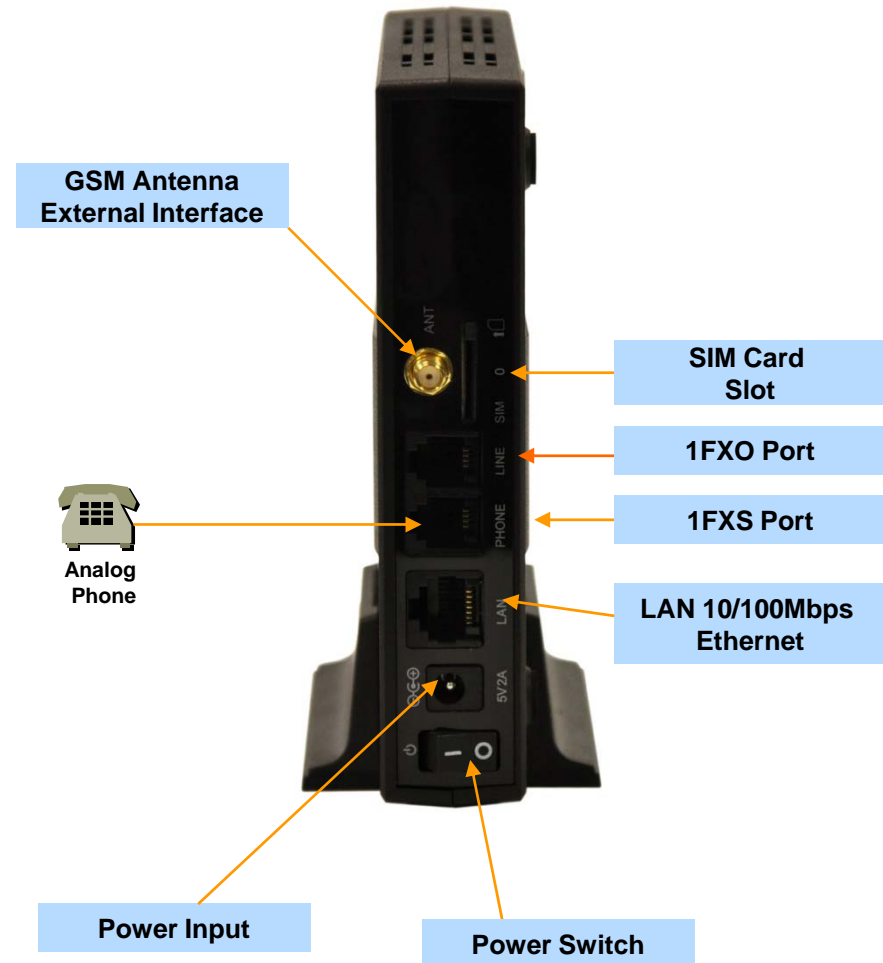
# Hardware Specification

## AP-GS701E 1-Port GSM Gateway

### Hardware Specifications

<b>AP-GS701E GSM Gateway</b>	<b>Basic Specifications</b>
<b>Voice Interface</b>	One(1)-Port FXS Analog Interface One(1)-Port FXO Analog Interface
<b>GSM Antenna Interface</b>	External GSM Antenna
<b>SIM Card Slot Interface</b>	GSM SIM Card Slot
<b>Ethernet Interface</b>	1-Port 10/100Mbps Ethernet Interface(RJ-45)
<b>Flash Memory</b>	4Mbyte High-speed Flash Memory
<b>Base Memory</b>	32 Mbyte High-speed SDRAM
<b>Power Requirement</b>	Power Supply Adaptor / VAC 110~220V, 50/60Hz, 5V, 2A
<b>Operating Temperature</b>	0°C ~ 45°C (32 °F ~ 122°F)
<b>Storage Temperature</b>	-40°C ~ 85°C (-40°C ~ 185°F)
<b>Relative Humidity</b>	5% ~ 95% (Non-condensing)
<b>Dimension</b>	142mm x 142mm x 28mm (W x D x H)

### Network interface Configurations





# GSM Module Specification

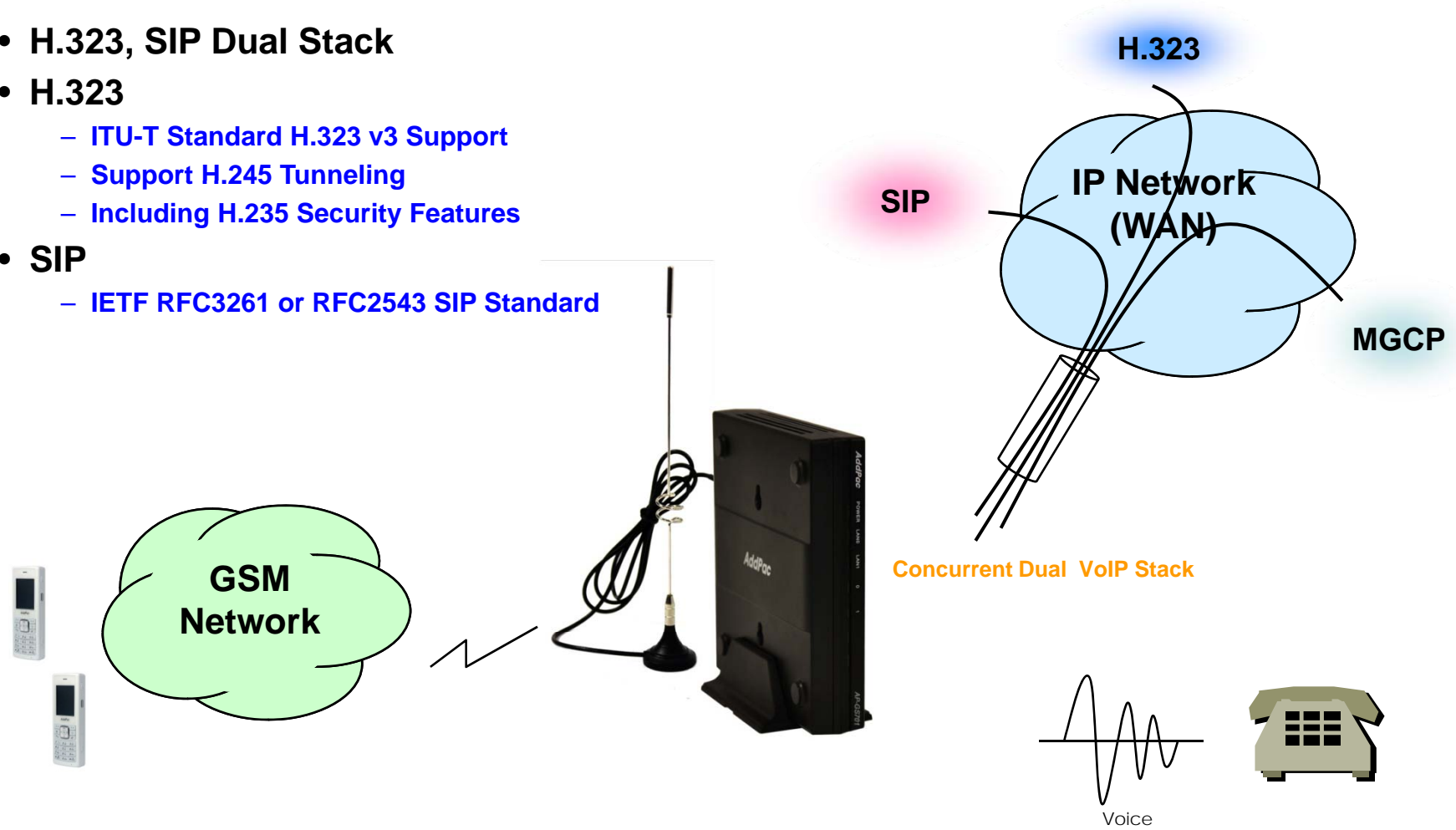
## AP-GS701E 1-Port GSM Gateway

- Bearers : GSM + GPRS multislots class 10
- Quad-Band EGSM 850/900/1800/1900MHz
- Normal Sensitivity
  - 850MHz Rx      -102 dBm (min.) -110 dBm (typ)
  - 900MHz Rx      -102 dBm (min.) -110 dBm (typ)
  - 1800MHz Rx     -102 dBm (min.) -109 dBm (typ)
  - 1900MHz Rx     -102 dBm (min.) -109 dBm (typ)
- Tx Performances
  - 850MHz Tx        +33 dBm (max.)
  - 900MHz Tx        +33 dBm (max.)
  - 1800MHz Tx       +30 dBm (max.)
  - 1900MHz Tx       +30 dBm (max.)
- Power Class 4 (33dBm nominal maximum output power)
- Codec
  - FR-EFR-HR-AMR

# VoIP (Voice over IP) Service

## AP-GS701E 1-Port GSM 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-GS701E 1-Port GSM 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-GS701E 1-Port GSM 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

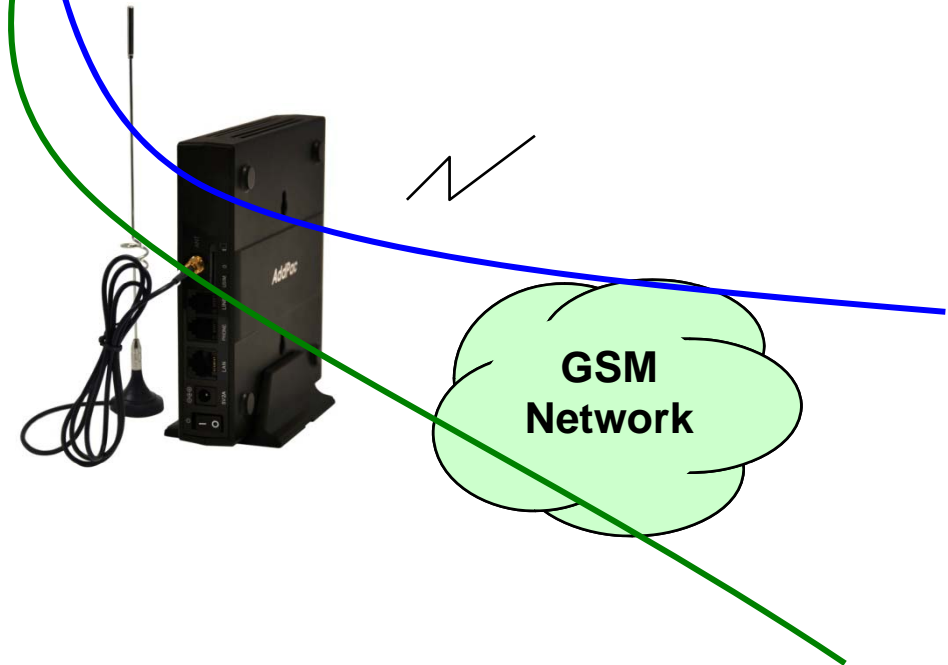
- **RTP**

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

VoIP



# VoIP (Voice over IP) Service

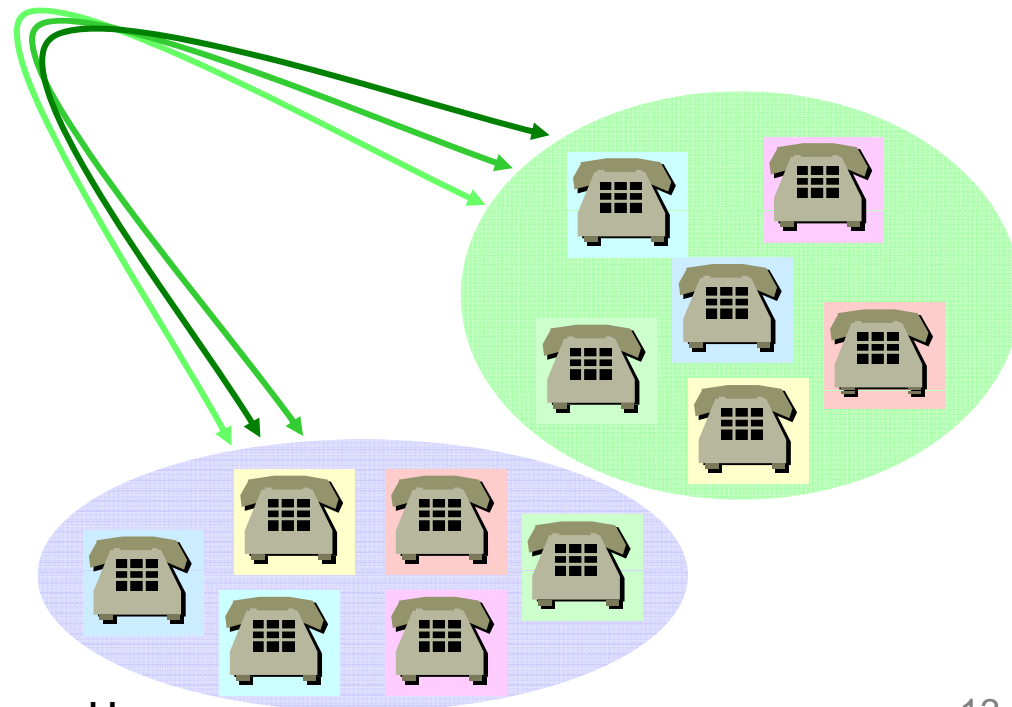
AP-GS701E 1-Port GSM Gateway

## • 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.)

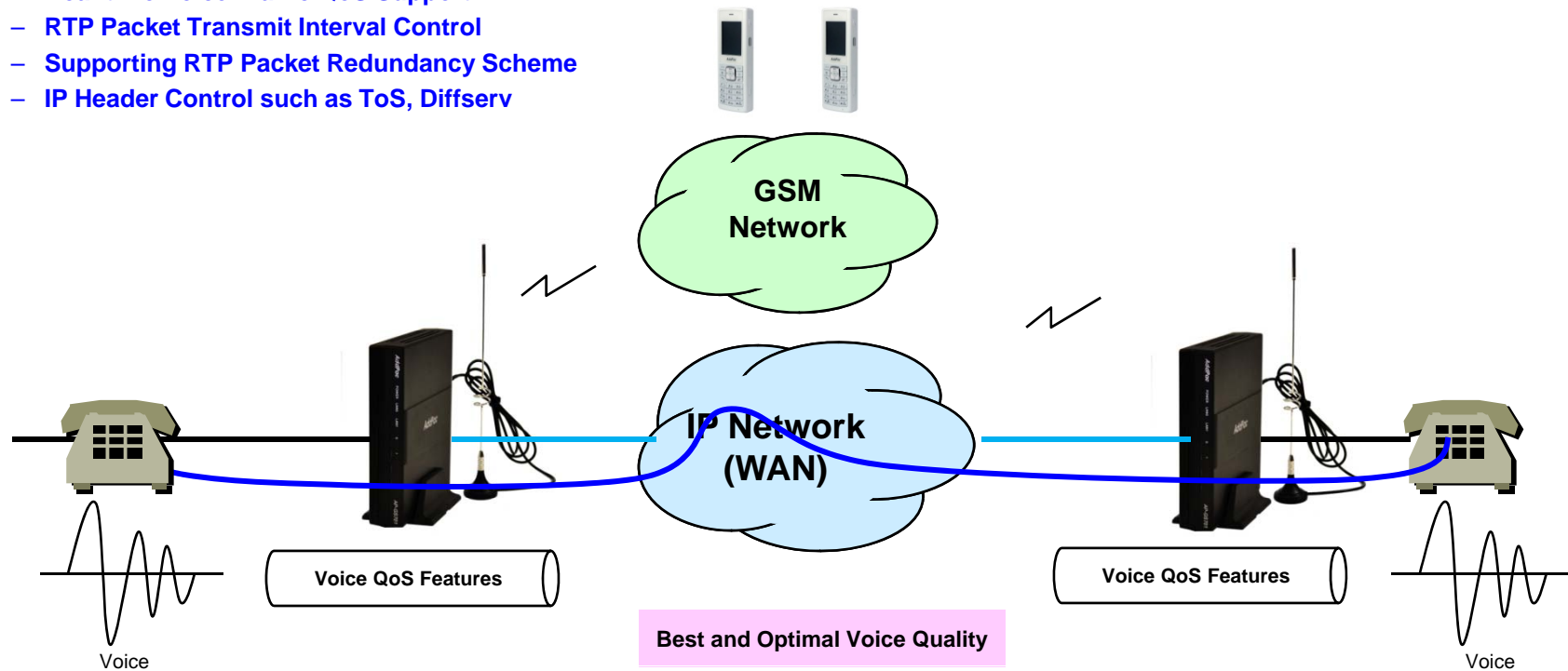
- Call transfer for internal calls
- Call pickup for internal calls
- Calling and called number conversion for VoIP outbound calls
- Calling and called number conversion for VoIP inbound calls
- Fax broadcasting call control



# Advanced QoS Features

## AP-GS701E 1-Port GSM 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-GS701E 1-Port GSM Gateway

## Basic Network Protocols

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

## Routing Protocol

- IPv4 : Static

## Service Protocol

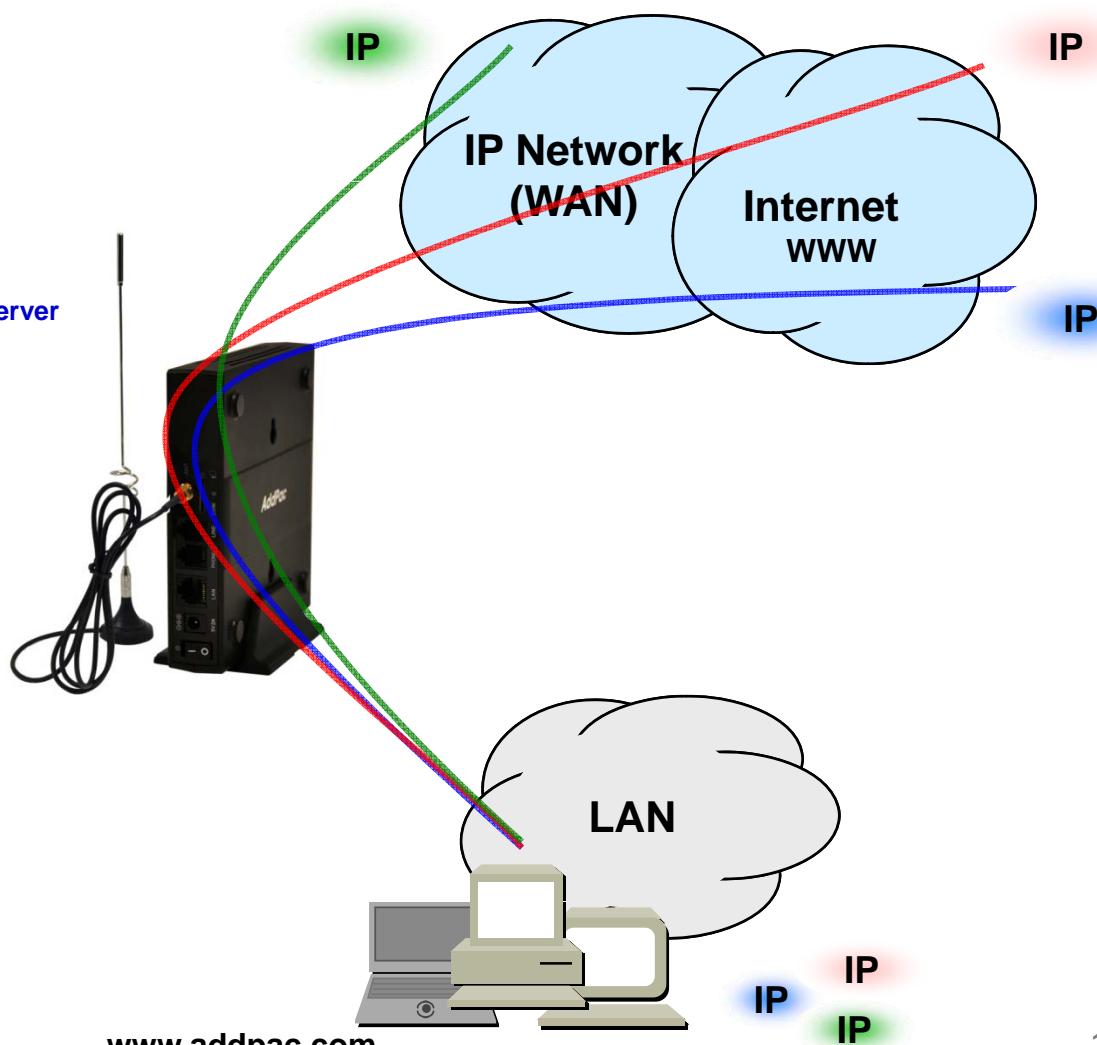
- FTP, Telnet, TFTP, DHCP Server/Relay, SNMP Server
- CDP (Cisco Discovery Protocol)
- DNS Resolver , DDNS(nsupdate)
- Bridge
- Syslog

## 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-GS701E 1-Port GSM 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**

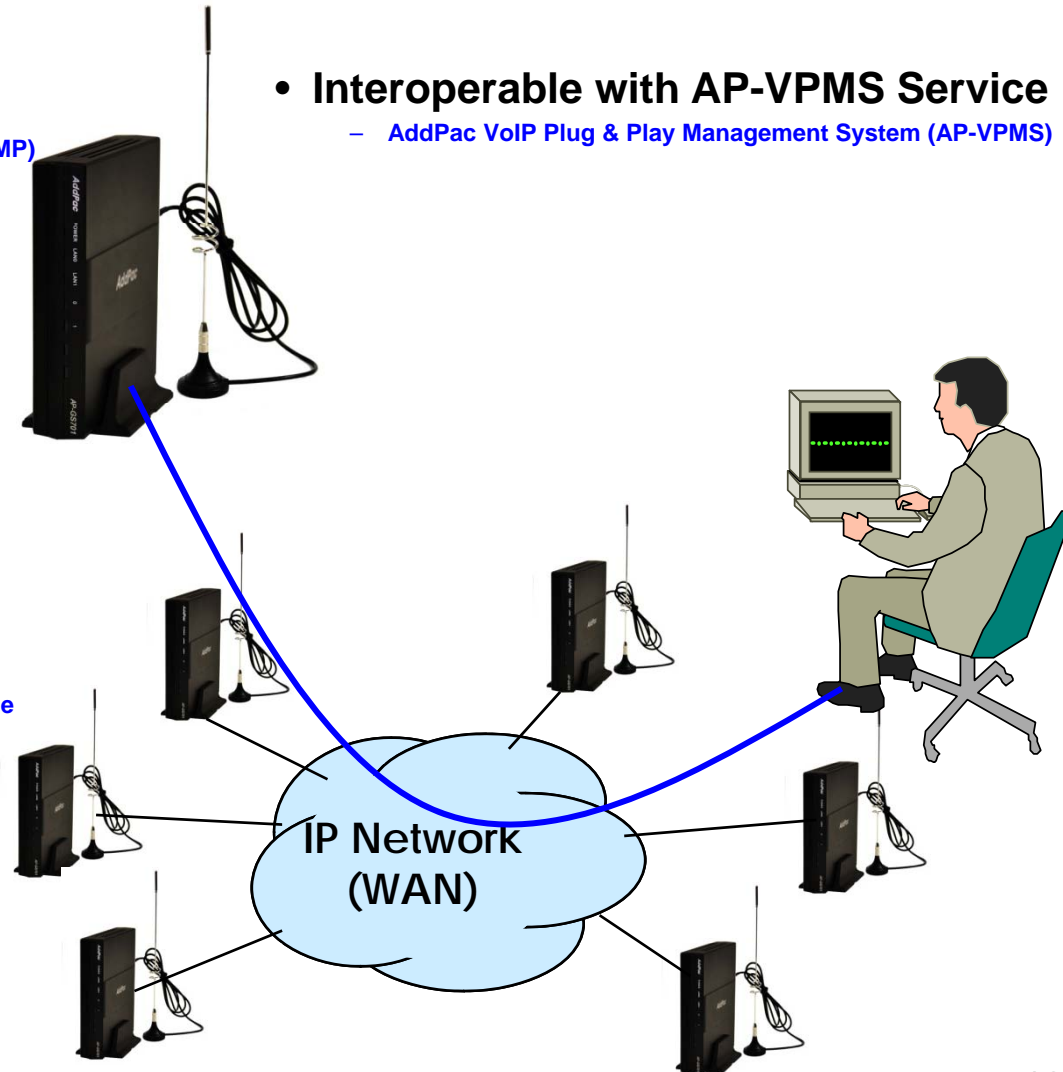
- HTTP server based APOS image and configuration file auto-upgrade support

- **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-GS701E 1-Port GSM 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

AddPac Tehonology  
Model : AP1800K\_G2  
H/W 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 않고사용자가 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)

## AP-GS701E 1-Port GSM Gateway

**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 : FXS Extension (Example)

## AP-GS701E 1-Port GSM Gateway

**Smart Web Manager**  
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**System**

- Network Setup
- Language
- NAT
- PPTP
- NTP

**Basic**

- Protocol
- Server SIP
- SIP Registration
- **FXS Extension**
- GSM Extension

**Advanced**

- Gain & CID
- GSM PINs
- Fax
- Service
- Filtering
- Security
- SNMP
- WEB Callback
- GSM Callback

**Miscellaneous**

- Call Status
- System Status
- Alarm Status
- GSM Status
- Call Log
- System Log
- Ping
- BTS Selection
- GSM BTS Info

### FXS Extension

**Port Information**

Port	P0	P1	P2	P3
SLOT0	GSM	GSM	FXS	FXS

**FXS Extension Configuration**

Index	Port	Numbers	Preference	HuntStop	Select
0	0/2	1234	0	0	<input type="checkbox"/>

**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**

Set up for using FXS port to extension number (forwarding No)

**Port Information**  
voice port type & physical port

**FXS Extension**  
Configure phone-number for using inter-office Preference ( 0 : highest )

# Smart Web Manager : GSM Extension (Example)

## AP-GS701E 1-Port GSM Gateway

The screenshot shows the Smart Web Manager interface for configuring GSM Extension. 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 configuring 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.

# Smart Web Manager : Call Status (Example)

## AP-GS701E 1-Port GSM Gateway

The screenshot shows the 'Call Status' page in the Smart Web Manager interface. The page is divided into several sections:

- System:** A sidebar menu with options like Network Setup, Language, NAT, DTMF, etc.
- Call Status:** The main content area, which includes:
  - Port Status (Analog):** A table showing the status of ports in Slot 0. The table has columns for Slot, Port, and Port Group (0(), 1(), 2(), 3()). The status for all ports is 'I' (Idle). Below the table are 'Unblock' and 'Block' buttons.
  - Call Status:** A table showing active call information. The table has columns for Port, Direction, Established Time, Calling Number, Called Number, CODEC, and Src/Dest. IP.
- Information:** A sidebar on the right containing system information such as Model (GS1002\_G2), H/W Version (2.0), S/W Version (8.00d), and Protocol (SIP).
- Description:** A sidebar on the right with the text: 'Verify port status and retrieve the present call information'.

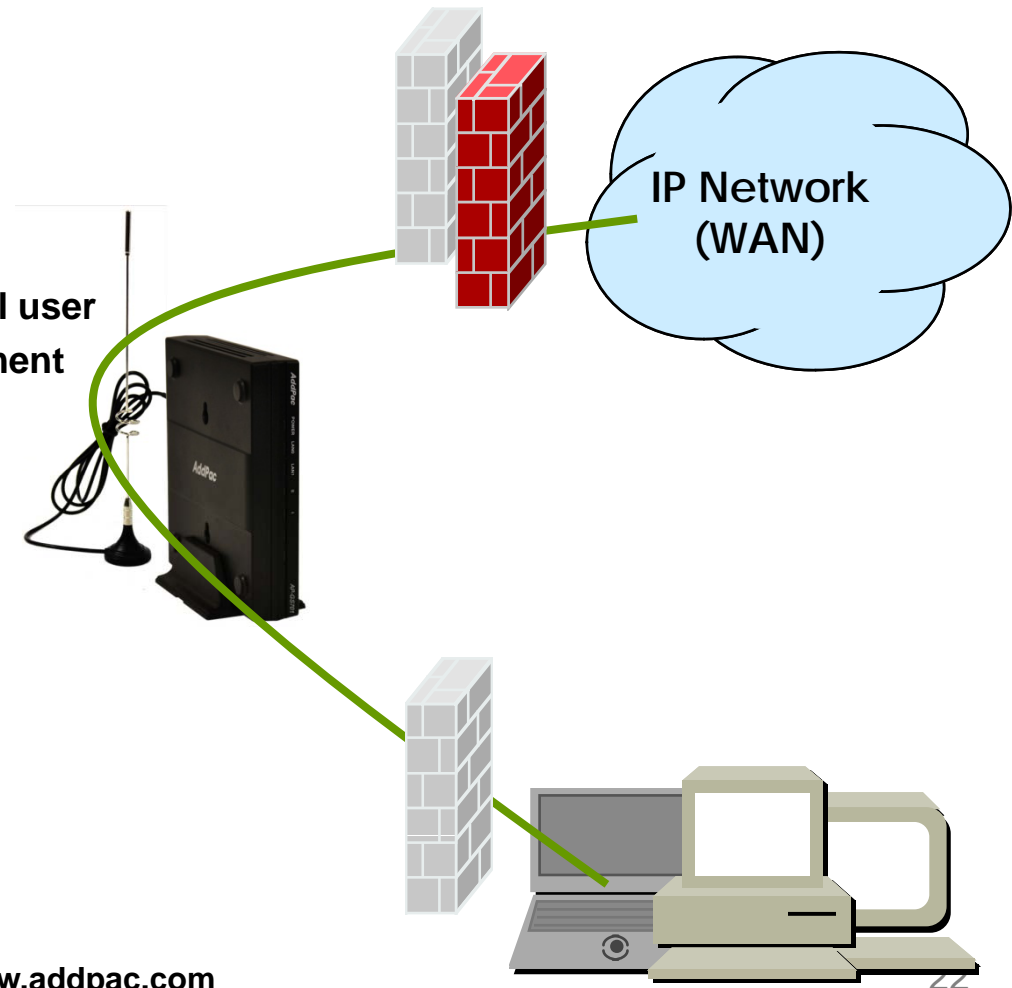
Two yellow callout boxes provide additional context:

- Analog Port:** Real-time display about analog port status (occupation, call status). Provide a specific port blocking function.
- Active Call Status:** Real-time display about current active call status (calling party addr, called party addr, Codec, etc).

# Security Management

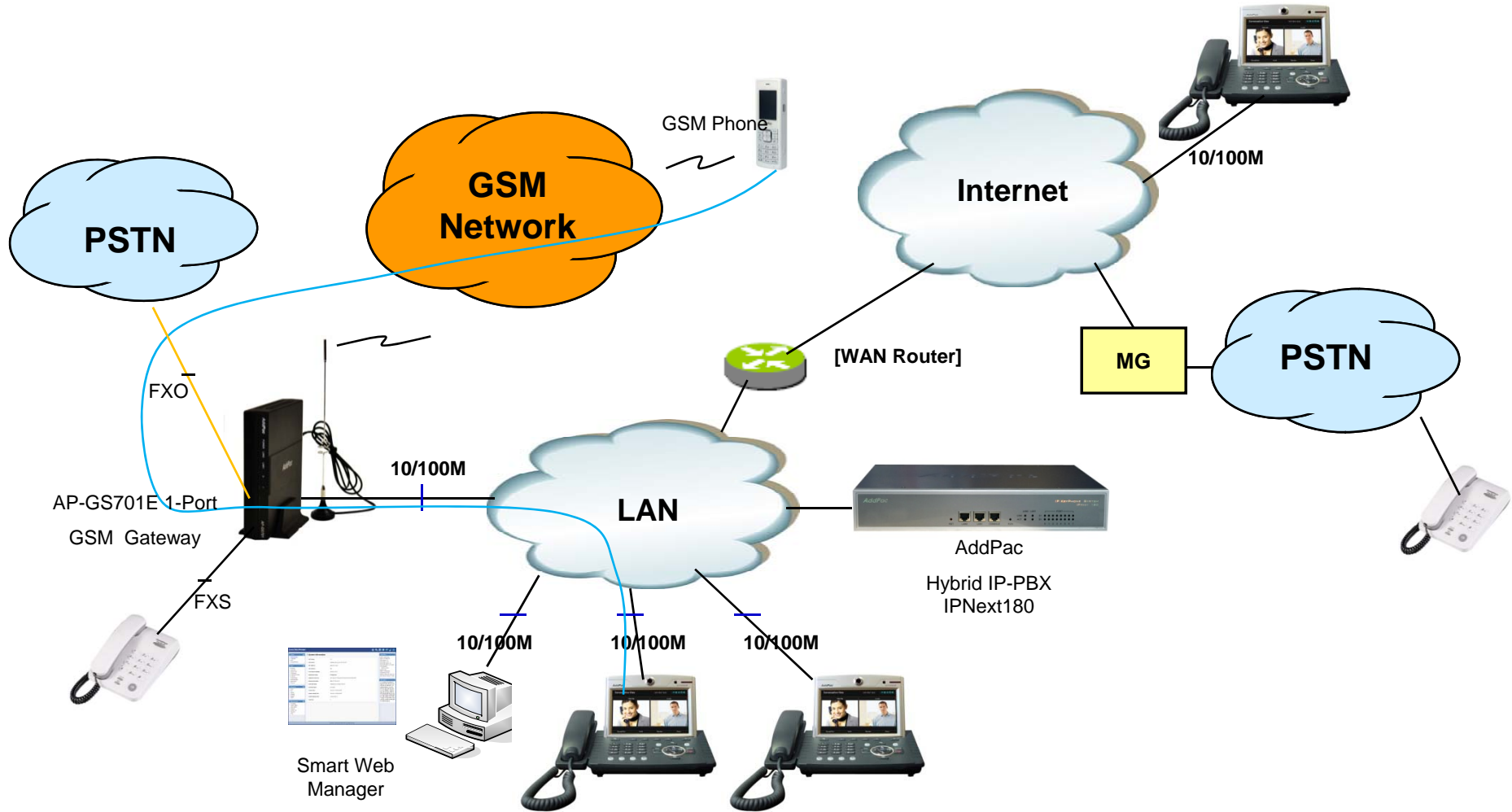
## AP-GS701E 1-Port GSM Gateway

- 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-GS701E 1-Port GSM Gateway



# Ordering Information

- AP-GS701E GSM Gateway Hardware
  - AP-GS701E GSM Gateway Main Body
  - RISC Microprocessor with High-end Programmable DSP Architecture
  - One(1) GSM Antenna & SIM Card Slot Interface
  - 1-Port FXS Interface(1xRJ11)
  - 1-Port FXO Interface(1xRJ11)
  - 1-Port 10/100Mbps Fast Ethernet(RJ45)
  - Including Network Cable, Antenna & Ext. Power Supply, etc.
- Built-in APOS Internetworking Software for AP-GS701E
- 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](http://www.addpac.com)



# GSM Gateway Series

# Thank you!

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