

# New 8-Port GSM VoIP Gateway Solution for Small & Medium Enterprise



**AddPac**

**AddPac Technology**

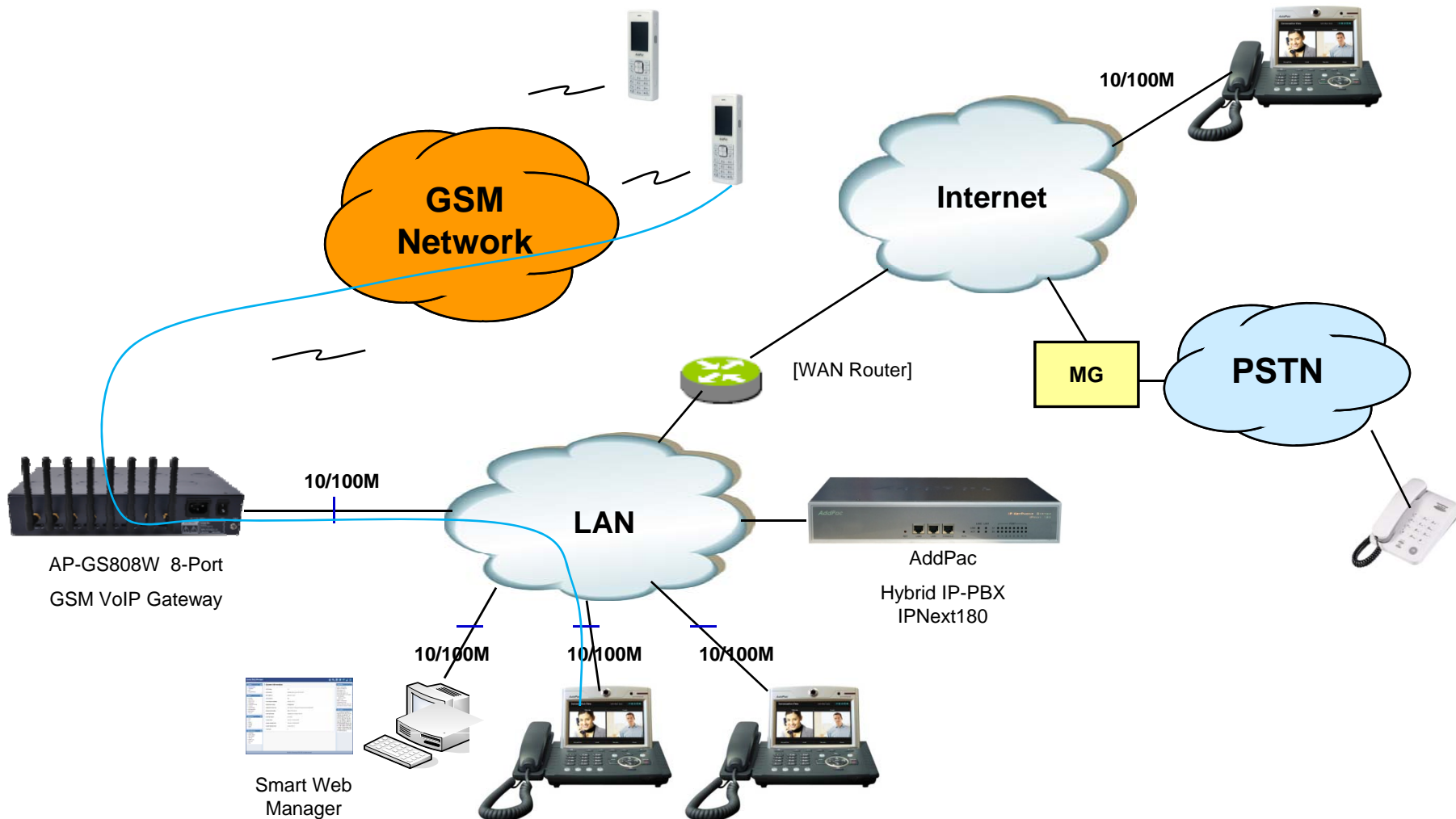
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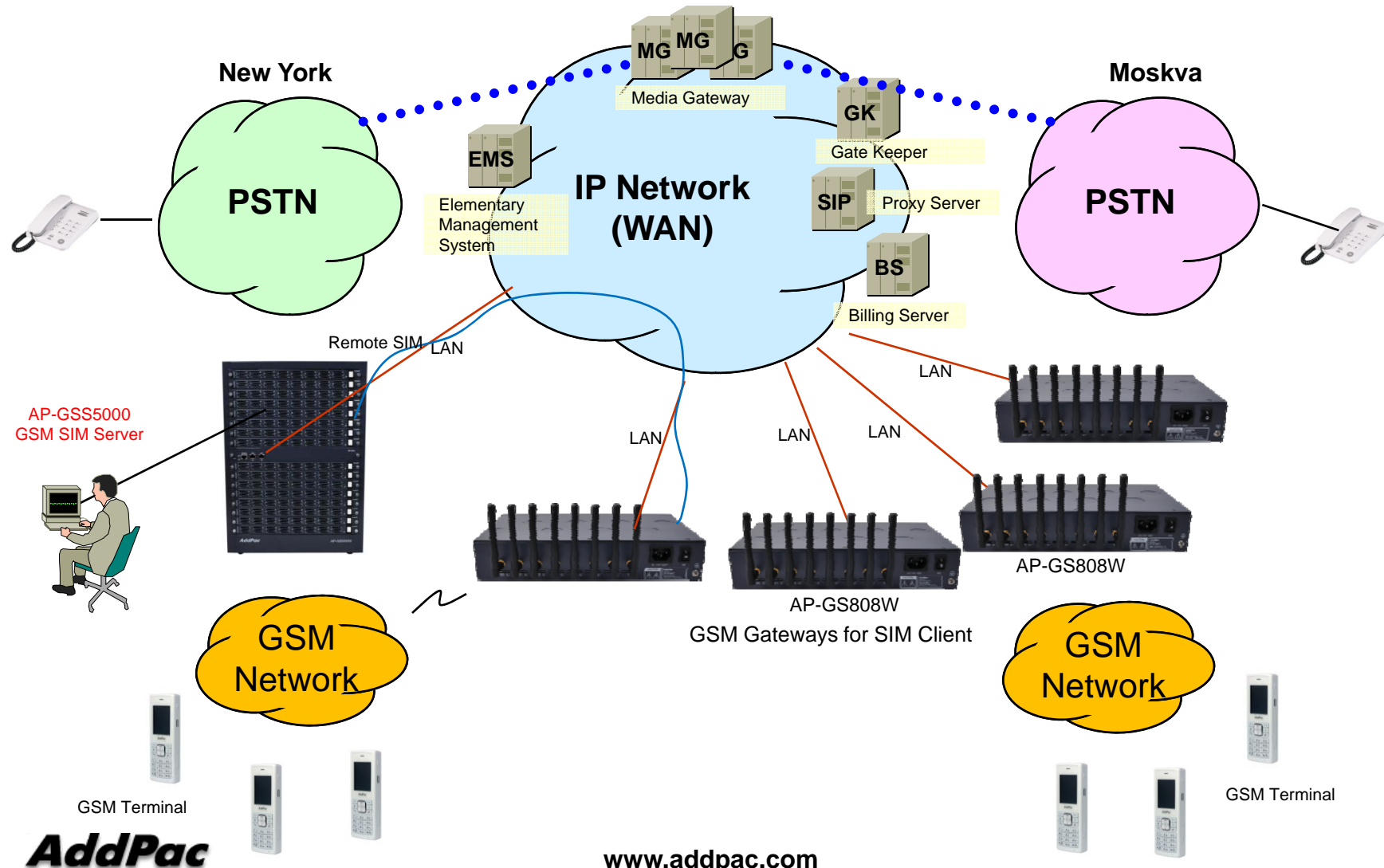
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  - AP-GS808
- Smart Web Manager for GSM Gateway

# Network Service Diagram (Standard App.)



# Network Service Diagram (GSM SIM Server)





# AP-GS808 8-Port GSM VoIP Gateway Series Comparison

# 8-Port GSM VoIP Gateway Comparison Table

| Model                                | AP-GS808W   | AP-GS808S   | AP-GS808  |
|--------------------------------------|---|---|---|
|                                      |  |  |  |
| GSM Module Vendor                    | Wavecom (Sierra)  | Wavecom (Sierra)  | UBlox   |
| GSM Channel                          | 8-Port  | 8-Port  | 8-Port  |
| IMEI change                          | Support :<br>Multiple Change  | Support :<br>Multiple Change  | Support :<br>One Time Change  |
| Auto BTS                             | Support   | N/A   | N/A   |
| SIM Client for Remote GSM SIM Server | Support   | N/A   | N/A   |
| External Antenna                     | 8 ea  | 8 ea  | 8 ea  |
| VoIP Codec                           | G.711, G.729, G.723.1,etc   | G.711, G.729, G.723.1,etc   | G.711, G.729, G.723.1,etc   |
| Power Supply                         | Internal Power  | Internal Power  | Internal Power  |



# AP-GS808W GSM VoIP Gateway

# Product Overview

## AP-GS808W 8-Port GSM VoIP Gateway

- 8-Port GSM VoIP Gateway Solution
- 8-Port GSM SIM Slot, 8-Port Antenna Interface
- GSM VoIP Interface(LAN) Support
- H.323/SIP Dual Concurrent Stack Embedded
- SIM Client Service for Remote SIM Application
- Auto IMEI change / Auto BTS Change
- 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 NMS for Large Scale Deployment
- Advanced Voice QoS Mechanism
- Internal Free Voltage Power Supply



# Hardware Specification

AP-GS808W 8-Port GSM VoIP Gateway

RISC  
CPU

High-end  
DSP

- RISC Microprocessor Computing Power
- Powerful High-End DSP for VoIP Interface
- 8-Port GSM VoIP Gateway
- 8-Port SIM Card Slots
- 8-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
- Internal Free Voltage Power Supply

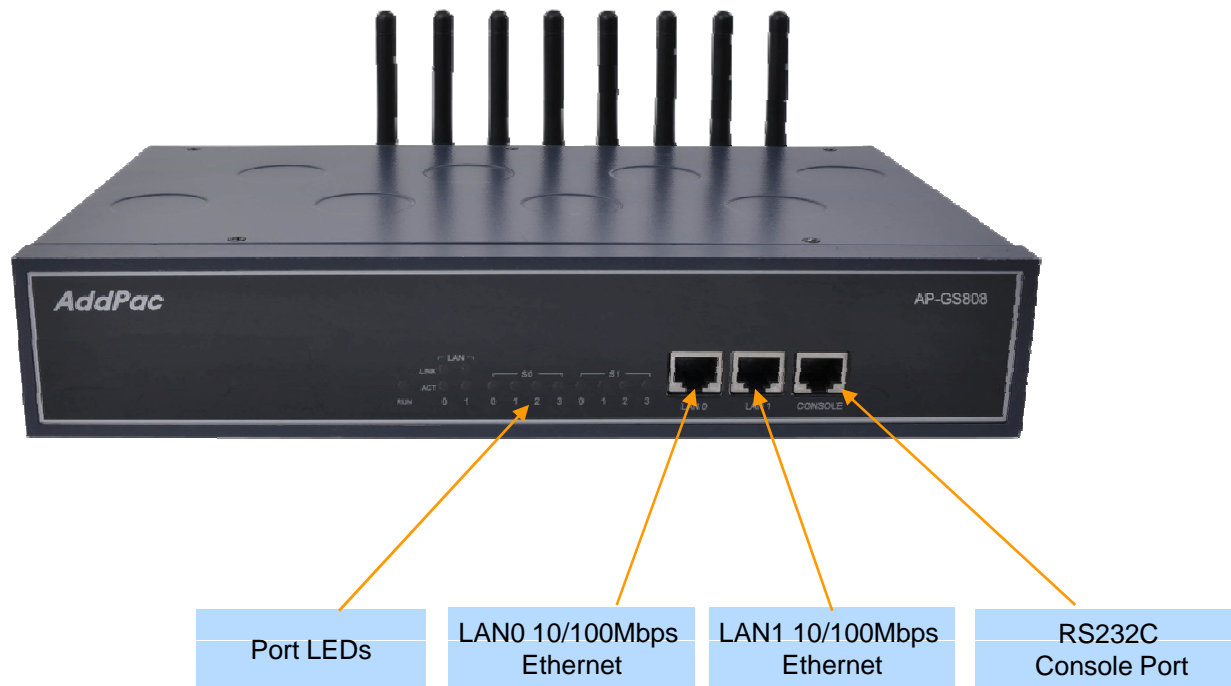
# Hardware Specification

AP-GS808W 8-Port GSM VoIP Gateway

RISC  
CPU

High-end  
DSP

## Front Side View



# Hardware Specification

AP-GS808W 8-Port GSM VoIP Gateway

RISC  
CPU

High-end  
DSP

## Back Side View





# AP-GS808S GSM VoIP Gateway

# Product Overview

## AP-GS808S 8-Port GSM VoIP Gateway

- 8-Port GSM VoIP Gateway Solution
- 8-Port GSM SIM Slot, 8-Port Antenna Interface
- GSM VoIP Interface(LAN) Support
- H.323/SIP Dual Concurrent Stack Embedded
- Auto IMEI change
- 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 NMS for Large Scale Deployment
- Advanced Voice QoS Mechanism
- Internal Free Voltage Power Supply

# Hardware Specification

AP-GS808S 8-Port GSM VoIP Gateway

RISC  
CPU

High-end  
DSP

- RISC Microprocessor Computing Power
- Powerful High-End DSP for VoIP Interface
- 8-Port GSM VoIP Gateway
- 8-Port SIM Card Slots
- 8-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
- Internal Free Voltage Power Supply





# AP-GS808 GSM VoIP Gateway

# Product Overview

## AP-GS808 8-Port GSM VoIP Gateway

- 8-Port GSM VoIP Gateway Solution
- 8-Port GSM SIM Slot, 8-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 NMS for Large Scale Deployment
- Advanced Voice QoS Mechanism
- Internal Free Voltage Power Supply



# Hardware Specification

## AP-GS808 8-Port GSM VoIP Gateway

RISC  
CPU

High-end  
DSP

- RISC Microprocessor Computing Power
- Powerful High-End DSP for VoIP Interface
- 8-Port GSM VoIP Gateway
- 8-Port SIM Card Slots
- 8-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
- Internal Free Voltage Power Supply





# Smart Web Manager for GSM Gateway

# Contents

- Main Page Layout
- System Configuration
  - Network Setup, Language, NAT, PPTP, NTP
- Basic Configuration
  - Protocol, SIP Server , FXS Extension, GSM Extension
  - DTMF/CODEC, VoIP Dial Plan, GSM Dial Plan, Static Routing, Hot Line
- Advanced Configuration
  - Gain/CID, GSM PINs, FAX, Service, Filtering, Security
  - GSM Web Callback, GSM Callback
- Miscellaneous Configuration
  - Call Status, System Status, Alarm Status, GSM Status
  - Call Log, System Log, Ping, BTS Selection, GSM BTS Info
- LCR(Least Cost Routing)
  - Black & White List, Time Interval, Tariff Group, LCR Test
- SMS
  - Inbox, SMS New Message

# Main Page Layout

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

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

**Information**  
Display the current system version and status summary

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

**Workspace**  
Workspace for detailed action

**System Information**

|                        |                             |
|------------------------|-----------------------------|
| H/W Version            | 2.0                         |
| SW Version             | 8.00d                       |
| MAC Address            | 0002.a400.0000              |
| VoIP Protocol          | SIP                         |
| Voice Interface Module | G(2)S(2)                    |
| Registration Status    | Registered                  |
| Supported Codec List   |                             |
| Network Information    | Static 172.16.9.16          |
| WAN LINK Status        | 100Mbps FULL Duplex Link UP |
| LAN LINK Status        | Link Down                   |
| Current Time           | Fri Jan 1 01:49:57 2010     |
| System Startup Time    | Fri Jan 1 00:00:00 2010     |
|                        | 0 days 01:49:57             |
|                        | 0                           |

**System**

- Network Setup
- Language
- NAT
- PPTP
- NTP

**Basic**

- Protocol
- Server SIP
- SIP Registration
- FXS Extension
- GSM Extension
- DTMF/CODEC
- VoIP Dial Plan
- GSM Dial Plan
- Static Route
- Hot Line

**Advanced**

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

**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 : Registered  
CurrentCalls: 0 Call  
Network : Static 172.16.9.16  
Mac Address: 0002.a400.0000  
Unread Message:  
P0:0(0)  
P0:1(0)

**Description**

# System – Network Setup

The screenshot shows the 'Network Setup' page in the Smart Web Manager. The page is divided into several sections, each with a radio button for selection. The 'Static IP' section is currently selected. The 'Information' panel on the right provides system details, and the 'Description' panel at the bottom right explains the WAN link configuration.

**Host Name**  
Create a representative name for the site to be installed

**Static IP**  
This is static IP mode. Specify the addressed IP from the service provider

**PPPoE**  
This is ADSL mode. This mode is used for addressing IP though authentication from the modem. At this time, the modem must be configured in a way that the device can be authenticated.

**DHCP**  
This is dynamic IP mode which is set at default. The IP can be addressed from the external DHCP server.

**VLAN**  
Configure VLAN mode and ID.

**WAN Link**  
Controls and recognizes WAN port. Specify the connection speed of WAN port. connection automatically.

**MAC**  
Change MAC address of WAN interface. Without address entry, use the basic MAC Address.

**Network Setup**

Hostname: GS1002

Static IP

IP Address: 172.16.9.16 (A.B.C.D)

Network Mask: 255.255.0.0 (A.B.C.D)

Default Router: 172.16.1.1 (A.B.C.D)

DNS Server: [ ] Primary DNS Server

[ ] Secondary DNS Server

PPPoE(ADSL)

Username: [ ]

Password: [ ]

DHCP

VLAN

ID: [0]

Auto

Speed:  100  10

Duplex:  full  half

Manual

MAC(Hardware) Address: [ ]-[ ]-[ ]-[ ]-[ ]-[ ]

Apply

**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 : Registered  
CurrentCalls : 0 Call  
Network : Static 172.16.9.16  
Mac Address : 0002.a400.0000  
Unread Message:  
P0:0(0)  
P0:1(0)

**Description**

This command sets up WAN port. The static IP address is to be assigned to the WAN port. The static IP address can be changed. MAC Address change can be used only when necessary. It is recommended to use the address created by the user not the address of the device

**UnRead SMS Messages**

# System - Language

**Smart Web Manager**  
www.addpac.com

**System**

- Network Setup
- NTP

**Basic**

- Protocol
- Server SIP
- SIP Registration
- FXS Extension
- GSM Extension
- DTMF/CODEC
- VoIP Dial Plan
- GSM Dial Plan
- Static Route
- Hot Line

**Advanced**

- Gain & CID
- GSM PINs

### Configure Language

한국어

English

Apply

**Information**

AddPac Technology  
Model : GS1002\_G2  
HW Version : 2.0  
SW Version : 8.00d  
Smart Web Version : 0.4  
Smart Web Build : Mar 24 2010  
Voice Interface  
G(2)S(2)  
Protocol : SIP  
Status : Registered  
CurrentCalls : 0 Call  
Network : Static 172.16.9.16  
Mac Address : 0002.a400.0000  
Unread Message:  
P0:0(0)  
P0:1(0)

**Description**

Choose the basic language to be applied. English is set at default.

**Configure Language**  
English, Korea

# System - NAT

**Smart Web Manager**  
www.addpac.com

**System**

- Network Setup
- Language
- **NAT**
- PPTP
- NTP

**Basic**

- DTMF/CODEC
- VoIP Dial Plan
- GSM Dial Plan
- Static Route
- Hot Line

**Advanced**

- Gain & CID
- GSM PINs
- Fax
- Service
- Filtering
- Security

## NAT Static Table

| IP Protocol | Global Port          | Local Address        | Local Port           | Control                            |
|-------------|----------------------|----------------------|----------------------|------------------------------------|
| tcp         | <input type="text"/> | <input type="text"/> | <input type="text"/> | <input type="button" value="Add"/> |

**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**

When many PCs are connected to LAN, create a table for delivering TCP/UDP port to PC.

**NAT Static Table**  
When many PCs are connected to LAN, create a table for delivering TCP/UDP port to PC

# System - PPTP

The screenshot shows the 'Smart Web Manager' interface for configuring PPTP. The main content area is titled 'Tunneling' and contains three sections: 'Mode', 'Source & Destination', and 'Service'. Three yellow callout boxes provide additional context:

- Tunneling Mode:** Configure Tunneling mode. This callout points to the 'Mode' section where 'None(Disable Tunneling, default)' is selected.
- Tunneling source & destination:** Source LAN port & destination IP. This callout points to the 'Source' dropdown (set to 'FastEthernet0/0') and the 'Destination' text input field.
- Tunneling Service:** Service mode. This callout points to the 'Service' section where 'Voice and Data Use Tunnel Interface (default)' is selected.

The 'Information' sidebar on the right provides system details:

- 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)

The 'Description' sidebar on the right states: 'This is the settings for using PPTP(Point-to-Point Tunneling Protocol). For the setting to be used for tunneling, please contact the manufacturer of the device'.



# System - NTP

**Smart Web Manager**  
www.addpac.com

**System**

- Network Setup
- Language
- NAT
- PPTP
- **NTP**

**Basic**

- Protocol
- Server SIP
- SIP Registration
- FXS Extension
- GSM Extension
- DTMF/CODEC
- VoIP Dial Plan
- GSM Dial Plan
- Static Route
- Utilities

**NTP**

Enable  Disable

Primary Server  (Domain Name or IP Address)

Secondary Server  (Domain Name or IP Address)

Interval  (1~72 hours)

Hours Offset  :  (-23~23 hours) : (0~60 minute)

**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**

**NTP**  
Configure NTP server (s) & Options

# Basic - Protocol

Smart Web Manager  
www.addpac.com

**System**

- Network Setup
- Language
- NAT
- PPTP
- NTP

**Basic**

- **Protocol**
- Server SIP
- SIP Registration
- FXS Extension
- GSM Extension
- DTMF/CODEC
- VoIP Dial Plan
- GSM Dial Plan
- Static Route
- Hot Line

**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

**Protocol**

SIP(Session Initiation Protocol)

Apply

**Configure VoIP signaling protocol SIP , H.323 (optional)**

**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 of the protocol to be used for VoIP communication

# Basic – SIP Server

**Smart Web Manager**  
www.addpac.com

**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

# Basic – FXS Extension

**Smart Web Manager**  
www.addpac.com

**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 )

# Basic – GSM Extension

**Smart Web Manager**  
www.addpac.com

**System**

- Network Setup
- Language
- NAT
- PPTP

**System**

- 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

**GSM Extension**

**Port Information**

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

**GSM Extension Configuration**

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

**GSM Extension with Translation**

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

**Information**

AddPac Technology  
Model : GS1002\_G2  
HW Version : 2.0  
SW 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 GSM port to extension number (forwarding No)

**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.

# Basic – DTMF/CODEC

**CODEC**  
Configure voice codec preference  
( g711a, g711u, g729, g7231, g726)

**DTMF**  
Configure DTMF relay method  
( in-band, RFC2833, out-of-band, CISCO type out-of-band)

**Smart Web Manager**  
www.addpac.com

**System**

- Network Setup
- Language

**Basic**

- Protocol
- Server SIP
- SIP Registration
- FXS Extension
- GSM Extension
- DTMF/CODEC**
- VoIP Dial Plan
- GSM Dial Plan
- Static Route
- Hot Line

**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

**DTMF/CODEC**

Voice CODEC

|              |      |
|--------------|------|
| Preference 1 | None |
| Preference 2 | None |
| Preference 3 | None |
| Preference 4 | None |
| Preference 5 | None |
| Preference 6 | None |

DTMF Relay mode

- DTMF relay by In-band voice
- DTMF relay by RTP payload defined by RFC 2833
- DTMF relay by Out-of-band signal
- DTMF relay by Cisco out-of-band signal

Apply

**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 GSM Dial Plan and Prefix table



# Basic – GSM Dial Plan

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

- Network Setup
- Language
- NAT
- PPTP
- NTP

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

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

**VoIP PLAN**  
Configure translation rule for GSM Peer.  
- first, 'Number of Digits to Delete' option is applied.  
- second, 'Digits to Insert' option is applied.

(ex) Origin called Number = 123456  
Number of Digits to Delete = 2  
Digits to Insert = "88"  
-----  
result = 883456

**Plan Table**

| Index | Digits to Insert | Number of Digits to Delete | Digit Pattern | Control                  |
|-------|------------------|----------------------------|---------------|--------------------------|
| 0     | 1                | 1                          | 2T            | <input type="checkbox"/> |

**Prefix Table**  
Configure GSM Peer with translation rule.

| Index | Prefix | 2nd Prefix | PlanIndex | Slot/Port | Control                  |
|-------|--------|------------|-----------|-----------|--------------------------|
| 0     | 33     | 2T         | 0         | 0/0       | <input type="checkbox"/> |

**Information**

AddPac Technology  
Model : GS1002\_G2  
HW Version : 2.0  
SW 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**

Hot Line Setup

**Miscellaneous**

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



# Basic – Static Route

**Smart Web Manager**  
www.addpac.com

**System**

- Network Setup
- Language
- NAT
- PPTP
- NTP

**Basic**

- Protocol
- Server SIP
- SIP Registration
- FXS Extension
- GSM Extension
- DTMF/CODEC
- VoIP Dial Plan
- GSM Dial Plan
- **Static Route**
- Hot Line

**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 RTS Info

### Static Route

Set Remote Site Call(5-digit number is set to begin \*2->\*2...)

| No | Remote Site IP       | Prefix               | Insert Digit         | Delete Digit         | Name of Remote Site  | Answer Addr          | Control                               |
|----|----------------------|----------------------|----------------------|----------------------|----------------------|----------------------|---------------------------------------|
| 0  | 172.16.1.1           | 2...                 | 172.16.9.16          | 0                    | Factory              | T                    | <input type="checkbox"/>              |
| *  | <input type="text"/> | <input type="text"/> | <input type="text"/> | <input type="text"/> | <input type="text"/> | <input type="text"/> | <input type="button" value="Delete"/> |
|    |                      |                      |                      |                      |                      |                      | <input type="button" value="Apply"/>  |

**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**

**Static Route**  
Configure Static VoIP Peer for using Inter-Office .  
( Already, I know IP & phone-number )

# Basic – Hot Line

**Smart Web Manager**  
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**Hot Line**

Hot Line Configuration

| Port    | Hot Line Number      | Digit Input Timeout <0~10 sec> |
|---------|----------------------|--------------------------------|
| S0P0(G) | <input type="text"/> | n.a.                           |
| S0P1(G) | <input type="text"/> | n.a.                           |
| S0P2(S) | 8888                 | 5                              |
| S0P3(S) | <input type="text"/> | <input type="text"/>           |

Apply

**Hot Line**  
- Used as baby-call(Connection PLAR)  
- Timer ( FXS port only : No Digit event is occurred for configured timer value, Auto-Dialing will be started )

**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**

# Advanced – Gain & CID

**Smart Web Manager**  
www.addpac.com

**System**

- Network Setup
- Language
- NAT
- PPTP
- NTP

**Basic**

- Protocol
- Server SIP
- SIP Registration
- FXS Extension
- GSM Extension
- DTMF/CODEC
- VoIP Dial Plan
- GSM Dial Plan
- Static Route
- Hot Line

**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 RTS Info

**Gain**

| Port | Port Type | InputGain | OutputGain | Caller ID                           |
|------|-----------|-----------|------------|-------------------------------------|
| P0:0 | GSM       | 0         | 0          | <input checked="" type="checkbox"/> |
| P0:1 | GSM       | 0         | 0          | <input checked="" type="checkbox"/> |
| P0:2 | FXS       | 0         | 0          | <input checked="" type="checkbox"/> |
| P0:3 | FXS       | 0         | 0          | <input checked="" type="checkbox"/> |

Apply

**Information**

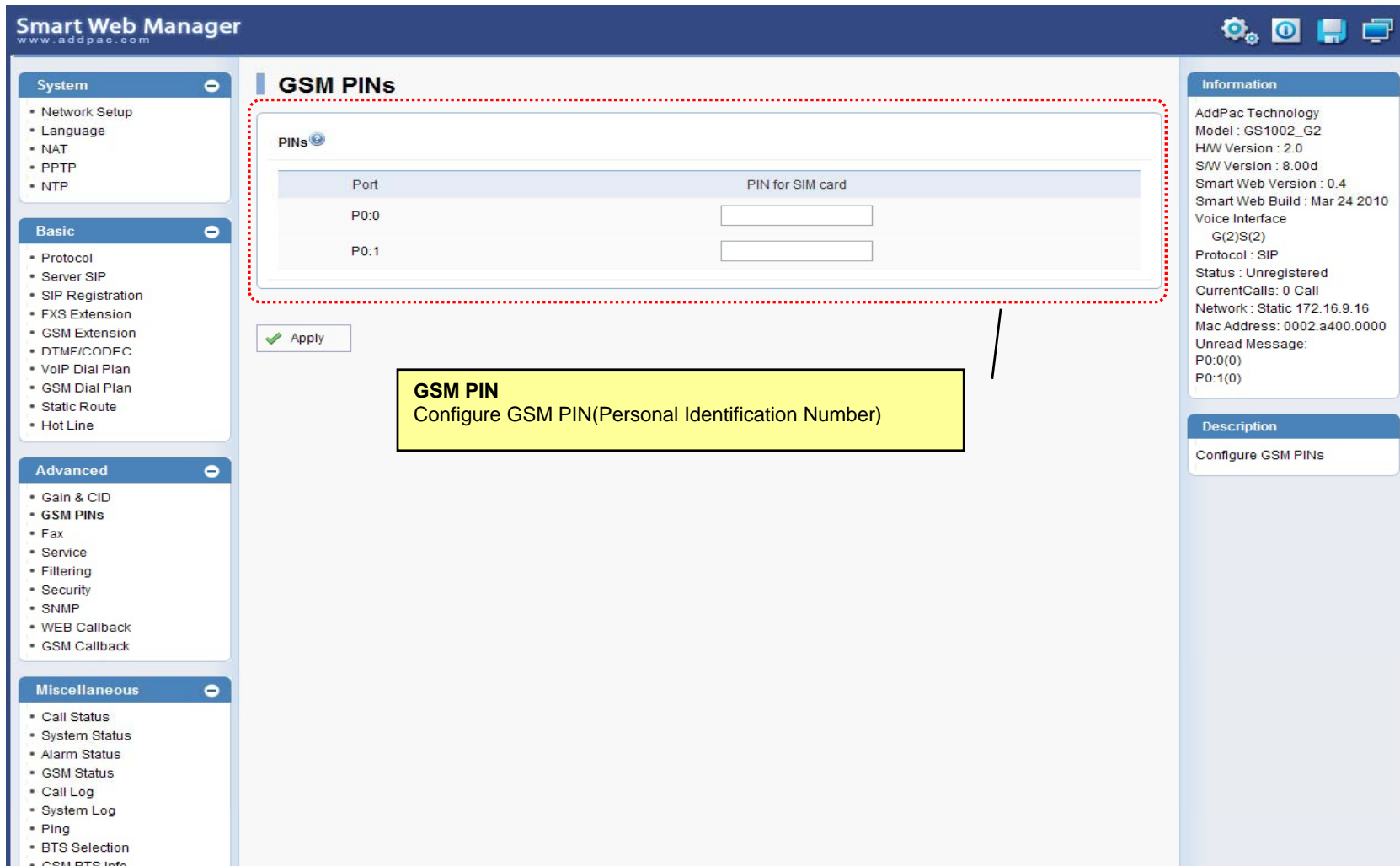
AddPac Technology  
Model : GS1002\_G2  
H/W Version : 2.0  
S/W Version : 8.00d  
Smart Web Version : 0.4  
Smart Web Build : Mar 24 2010  
Voice Interface  
G(2)S(2)  
Protocol : SIP  
Status : Unregistered  
CurrentCalls: 0 Call  
Network : Static 172.16.9.16  
Mac Address: 0002.a400.0000  
Unread Message:  
P0:0(0)  
P0:1(0)

**Description**

Adjust the input voice volume from FXS/FXO/E1/E&M to DSP and the output volume from DSP to the phone or PSTN line;

**Gain & CID**  
Configure Input-gain, output-gain and caller-ID.

# Advanced – GSM PINs



The screenshot shows the Smart Web Manager interface for configuring GSM PINs. The main content area is titled "GSM PINs" and contains a table with two columns: "Port" and "PIN for SIM card". The table has two rows: "P0:0" and "P0:1", each with an empty input field for the PIN. A red dashed box highlights the table. Below the table is an "Apply" button with a green checkmark icon. A yellow callout box with a black border points to the table and contains the text: "GSM PIN Configure GSM PIN(Personal Identification Number)".

| Port | PIN for SIM card     |
|------|----------------------|
| P0:0 | <input type="text"/> |
| P0:1 | <input type="text"/> |

Apply

**GSM PIN**  
Configure GSM PIN(Personal Identification Number)

**System**

- Network Setup
- Language
- NAT
- PPTP
- NTP

**Basic**

- Protocol
- Server SIP
- SIP Registration
- FXS Extension
- GSM Extension
- DTMF/CODEC
- VoIP Dial Plan
- GSM Dial Plan
- Static Route
- Hot Line

**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

**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 GSM PINs

# Advanced - Fax

**Smart Web Manager**  
www.addpac.com

**System**

- Network Setup
- Language
- NAT
- PPTP
- NTP

**Basic**

- Protocol
- Server SIP
- SIP Registration
- FXS Extension
- GSM Extension
- DTMF/CODEC
- VoIP Dial Plan
- GSM Dial Plan
- Static Route
- Hot Line

**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

**Fax**

**Fax Mode**  T.38  Inband T.38  Bypass

**Fax Rate**  Disable  2400  4800  7200  9600  12000  14400

Apply

**FAX**  
Configure fax mode & rate (VoIP Lines)

**Information**

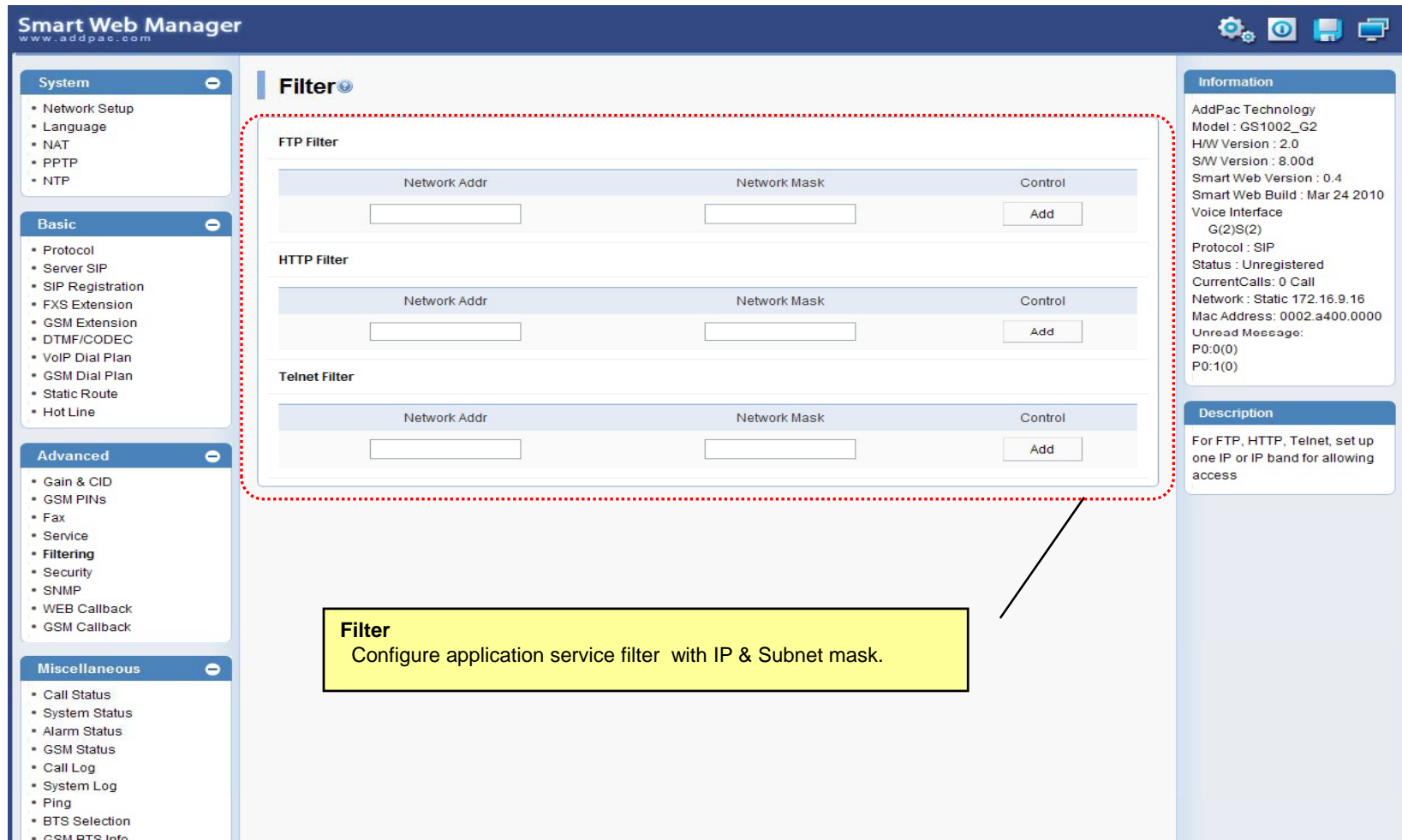
AddPac Technology  
Model : GS1002\_G2  
H/W Version : 2.0  
S/W Version : 8.00d  
Smart Web Version : 0.4  
Smart Web Build : Mar 24 2010  
Voice Interface  
G(2)S(2)  
Protocol : SIP  
Status : Unregistered  
CurrentCalls: 0 Call  
Network : Static 172.16.9.16  
Mac Address: 0002.a400.0000  
Unread Message:  
P0:0(0)  
P0:1(0)

**Description**

Enable or disable T.38/Inband T.38, which is fax internet protocol and specify Baudrate



# Advanced - Filtering



**Smart Web Manager**  
www.addpac.com

**System**

- Network Setup
- Language
- NAT
- PPTP
- NTP

**Basic**

- Protocol
- Server SIP
- SIP Registration
- FXS Extension
- GSM Extension
- DTMF/CODEC
- VoIP Dial Plan
- GSM Dial Plan
- Static Route
- Hot Line

**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

**Filter**

**FTP Filter**

| Network Addr         | Network Mask         | Control                            |
|----------------------|----------------------|------------------------------------|
| <input type="text"/> | <input type="text"/> | <input type="button" value="Add"/> |

**HTTP Filter**

| Network Addr         | Network Mask         | Control                            |
|----------------------|----------------------|------------------------------------|
| <input type="text"/> | <input type="text"/> | <input type="button" value="Add"/> |

**Telnet Filter**

| Network Addr         | Network Mask         | Control                            |
|----------------------|----------------------|------------------------------------|
| <input type="text"/> | <input type="text"/> | <input type="button" value="Add"/> |

**Information**

AddPac Technology  
Model : GS1002\_G2  
HW Version : 2.0  
SW 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**

For FTP, HTTP, Telnet, set up one IP or IP band for allowing access

**Filter**  
Configure application service filter with IP & Subnet mask.

# Advanced - Security

**Smart Web Manager**  
www.addpac.com

**System**

- Network Setup
- Language
- NAT
- PPTP
- NTP

**Basic**

- Protocol
- Server SIP
- SIP Registration
- FXS Extension
- GSM Extension
- DTMF/CODEC
- VoIP Dial Plan
- GSM Dial Plan
- Static Route
- Hot Line

**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

**Security**

IP Filtering  Enable  Disable

WarDialing Filtering  Enable  Disable

Allow Digit Length(IP to PSTN) Min  Max

SIP Shutdown  Enable  Disable

Apply

**Information**

AddPac Technology  
Model : GS1002\_G2  
HW Version : 2.0  
SW 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 IP authentication and protocol and incoming number

**Security**

- IP Filtering : Allowing only the inbound call which is registered to Call-Routing of the server by static IP.
- WarDialing : Allowing only the inbound call with the number registered to Inter-Office and phone-number.
- Digit Length : Allowing only the inbound call with the number registered to Inter-Office and phone-number
- SIP : SIP signaling packets are filtered.



# Advanced – GSM Web Callback

**Smart Web Manager**  
www.addpac.com

**System**

- Network Setup
- Language
- NAT
- PPTP
- NTP

**Basic**

- GSM Extension
- DTMF/CODEC
- VoIP Dial Plan
- GSM Dial Plan
- Static Route
- Hot Line

**GSM WEB Callback**

**Calling Number Whitelist**

| index | DialPattern | Control                  |
|-------|-------------|--------------------------|
| 0     | 01023234444 | <input type="checkbox"/> |

0 [input type="text"/> [input type="button" value="Delete"/> [input type="button" value="Add"/>

**WEB Callback**

Destination Numbers [input type="text" value="8888"/> Source Numbers [input type="text" value="9999"/> [input type="button" value="Apply"/>

**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**

Execute GSM call service function and Configure call service whitelist

**Calling Number White List**  
The employee working at the out of office are usually registered.

**WEB Callback**  
The employee working at the out of office can use web call agent.

**Status Viewer**  
Display real-time call status.

# Advanced – GSM Callback

**GSM Callback**  
 The employee working at the out of office can use this function.

- The Call received from GSM network is automatically disconnected.
- GSM gateway calls to the calling number.

**Calling Number White List**  
 The employee working at the out of office are usually registered.

**Smart Web Manager**  
www.addpac.com

**GSM Callback**

**Calling Number Whitelist**

| Group | Index | DialPattern | Control                  |
|-------|-------|-------------|--------------------------|
| 3     | 0     | 123T        | <input type="checkbox"/> |

3 | 0 | [ ] | [Delete] | [Add]

**Callback**

| GSM Port | My Number | WhiteList Group |
|----------|-----------|-----------------|
| P0:0     |           | 3               |
| P0:1     |           | N.A.            |

[Apply]

**Information**

AddPac Technology  
 Model : GS1002\_G2  
 HW Version : 2.0  
 SW 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**

Execute GSM callback function and Configure callback whitelist

**Callback**  
 The white list group is adapted to specific GSM port

# Miscellaneous – Call Status

**Smart Web Manager**  
www.addpac.com

**System**

- Network Setup
- Language
- NAT
- DTMF/CODEC
- VoIP Dial Plan
- GSM Dial Plan
- Static Route
- Hot Line

**Miscellaneous**

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

**Call Status**

**Port Status (Analog)**

| Slot   | Port  | Port Group               |                          |                          |                          |
|--------|---|--------------------------|--------------------------|--------------------------|--------------------------|
|        |   | 0()                      | 1()                      | 2()                      | 3()                      |
| SLOT 0 | Status  | I                        | I                        | I                        | I                        |
|        | Select  | <input type="checkbox"/> | <input type="checkbox"/> | <input type="checkbox"/> | <input type="checkbox"/> |
|        | <input type="button" value="Unblock"/> <input type="button" value="Block"/> |                          |                          |                          |                          |

Connection State :  (Connected)  (Disconnected || Blocked)

Call State :  (Idle)  (Ring || Dial)  (Called)  (Calling)  (Blocked)

**Call Status**

| Port | Direction | Established Time | Calling Number | Called Number | CODEC | Src/Dest. IP |
|------|-----------|------------------|----------------|---------------|-------|--------------|
|      |           |                  |                |               |       |              |

**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**

Verify port status and retrieve the present call information

**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)

# Miscellaneous – System Status

**System Status**

- voice port status & information
- SIP-UA status & information
- gateway status & information
- system utilization information

Smart Web Manager  
www.addpac.com

**System**

- Network Setup
- Language
- NAT
- PPTP
- NTP

**System Status**

**Voice Port**

| Port | LineType | Status | InGain | OutGain | TieType  | TieDigits | CallNum | Tcalled | Tcalling |
|------|----------|--------|--------|---------|----------|-----------|---------|---------|----------|
| 0/ 0 | GSM      | Idle   | 0      | 0       | none     |           | -1      | -1      | -1       |
| 0/ 1 | GSM      | Idle   | 0      | 0       | none     |           | -1      | -1      | -1       |
| 0/ 2 | FXS      | Idle   | 0      | 0       | hot-line | 8888      | -1      | -1      | -1       |
| 0/ 3 | FXS      | Idle   | 0      | 0       | none     |           | -1      | -1      | -1       |

**SIP-UA**

Proxyserver Registration Information  
proxyserver registration option = e164  
Proxyserver list :

| Server address | Port | Priority | Domain | Status (LastFailReason) |
|----------------|------|----------|--------|-------------------------|
| 172.17.116.215 | 5060 | 128      | any    | Failed(Rx:OtherMsg)     |

Proxy Server registration status :

| E.164 | UserName | Password | Port | Status     |
|-------|----------|----------|------|------------|
| 1005  | 1005     | NONE     | 0/ 2 | Registered |
| 33    | 33       | NONE     | 0/ 0 | Failed     |

SIP UA Timer counters  
retry counter = 10

SIP UA Timer values  
 tretry (sip retry timer) = 500 msec.  
 tinterval (sip retry max interval timer) = 4 sec.  
 treg (sip register timer) = 60 sec.  
 tregtry (sip register retry timer) = 20 sec.  
 texpires (sip invite expire timer) = 180 sec.  
 tsipping (sip ping timer) = 45 sec.

SIP UA Session Timer value  
 Min-SE = 1800 sec.  
 Session-Expires = 1800 sec.

SIP DNS SRV Query : Disable  
 SIP Call Transfer Mode : Basic  
 SIP Media Channel Start Mode : Default  
 SIP Reliable Provisional Response Option : Supported with value <100rel>  
 SIP Response Option : default  
 SIP Local Domain : NULL  
 SIP Special Char : NULL  
 SIP Routing Method of Incoming Call : Default  
 SIP Remote-Party-ID : Disabled

**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**

Verify the present port information, Server Register status, CPU and Memory usage

**Advanced**

- VoIP Dial Plan
- GSM Dial Plan
- Static Route
- Hot Line

**Miscellaneous**

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

**Call Status**

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

# Miscellaneous – Alarm Status

**Smart Web Manager**  
www.addpac.com

**System**

- Network Setup
- Language
- NAT
- PPTP
- NTP

**Basic**

- Protocol
- Server SIP
- SIP Registration
- FXS Extension
- GSM Extension
- DTMF/CODEC
- VoIP Dial Plan
- GSM Dial Plan
- Static Route
- Hot Line

**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

**Alarm Status**

Interface Down Counter = 0  
H323 Register Fail Counter = 0  
SIP Register Fail Counter = 0

Clear

**Alarm Status**

- Interface Down count
- H323 register fail counter
- SIP register fail counter

**Information**

AddPac Technology  
Model : GS1002\_G2  
HW Version : 2.0  
SW 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**

Verify the physical failure of LAN, H.323 / SIP Register Fail by increasing counter

# Miscellaneous – GSM Status

**Smart Web Manager**  
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**GSM Status**

**GSM Port Status & Information**

| Port | My Phone Number | Device Information |                 | Accounting (Used/Quota/Free) |                 |
|------|-----------------|--------------------|-----------------|------------------------------|-----------------|
|      |                 | Register Status    | Signal Strength | Voice Quota(secs)            | SMS Quota(E.A.) |
| P0:0 |                 | REG                | 0dB             | 0 / 36000 / 36000            | 0 / 300 / 300   |
| P0:1 |                 | REG                | 0dB             | 0 / -1 / -1                  | 0 / -1 / -1     |

**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)

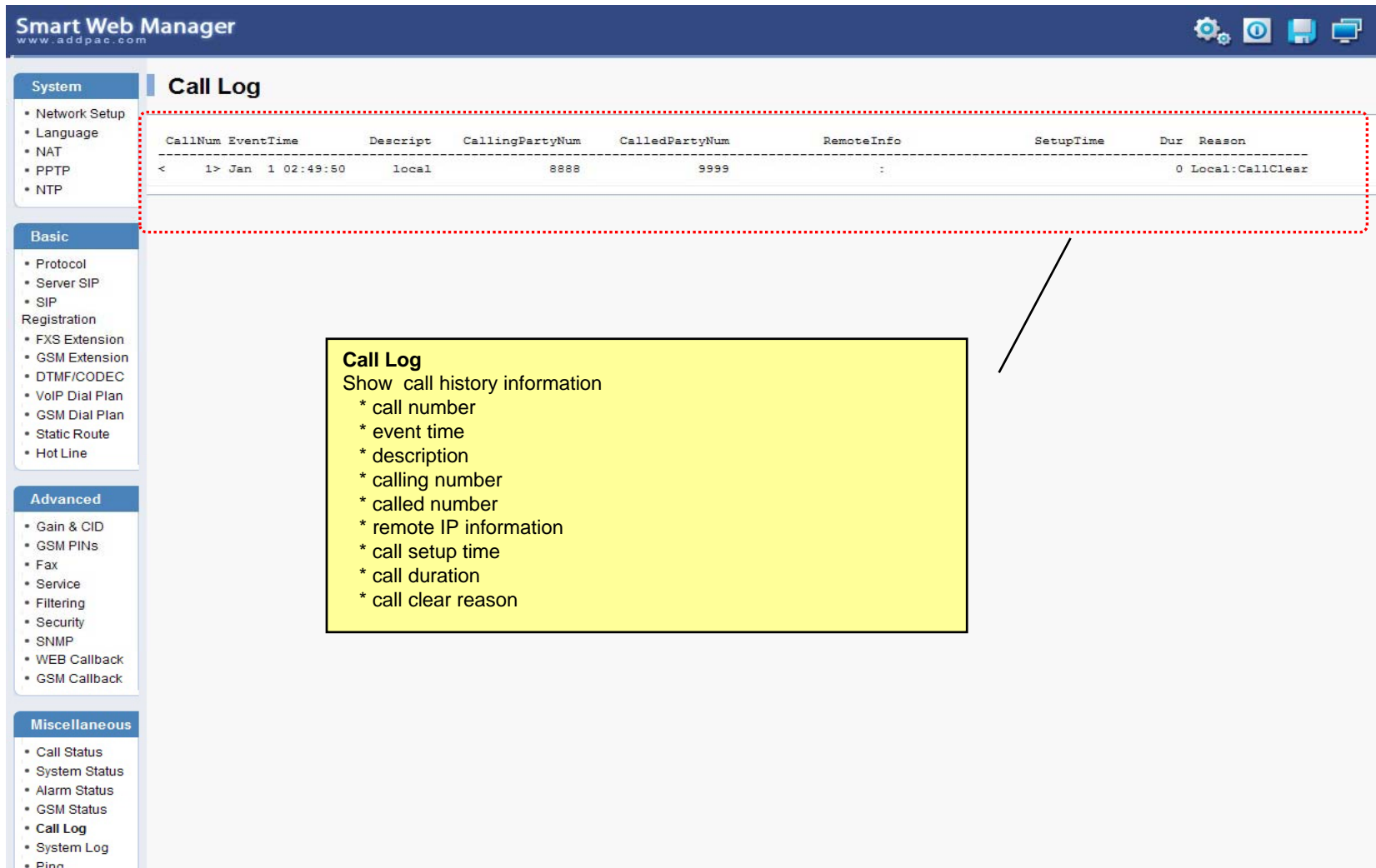
**Description**

Diaplay GSM wireless status

**GSM Status**

- my number
- GSM register status
- GSM signal strength
- Account information
  - \* voice quota ( used / quota / free )
  - \* SMS quota ( used / quota / free )

# Miscellaneous – Call Log



The screenshot shows the 'Smart Web Manager' interface with the 'Call Log' section active. The interface includes a left sidebar with navigation menus for System, Basic, Advanced, and Miscellaneous. The main content area displays a table of call log entries. A red dashed box highlights the table header and the first entry. A yellow callout box provides a list of fields shown in the table.

| CallNum | EventTime      | Descript | CallingPartyNum | CalledPartyNum | RemoteInfo | SetupTime | Dur | Reason          |
|---------|----------------|----------|-----------------|----------------|------------|-----------|-----|-----------------|
| < 1>    | Jan 1 02:49:50 | local    | 8888            | 9999           | :          |           | 0   | Local:CallClear |

**Call Log**  
Show call history information

- \* call number
- \* event time
- \* description
- \* calling number
- \* called number
- \* remote IP information
- \* call setup time
- \* call duration
- \* call clear reason

# Miscellaneous – System Log

**Smart Web Manager**  
www.addpac.com

**System Log**

command logging buffers (messages logged)  
-----  
event logging buffers (messages logged)  
-----

**System Log**  
- command log  
- system alarm log ( ex : interface down)

**Information**  
AddPac Technology  
Model : GS1002\_G2  
HW Version : 2.0  
SW 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**  
Retrieve the system log

**System**  
• Network Setup  
• Language  
• NAT  
• PPTP  
• NTP

**Basic**  
• Protocol  
• Server SIP  
• SIP Registration  
• FXS Extension  
• GSM Extension  
• DTMF/CODEC  
• VoIP Dial Plan  
• GSM Dial Plan  
• Static Route  
• Hot Line

**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



# Miscellaneous - Ping

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

- Network Setup
- Language
- NAT

**Advanced**

- Protocol
- Server SIP
- SIP Registration
- FXS Extension
- GSM Extension
- DTMF/CODEC
- VoIP Dial Plan
- GSM Dial Plan
- Static Route
- Hot Line

**Miscellaneous**

- Call Status
- System Status
- [Alarm Status](#)
- GSM Status
- Call Log
- System Log
- **Ping**
- BTS Selection
- GSM BTS Info

**Ping**

Host address  Start

**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**

Verify the physical failure of LAN, H.323 / SIP Register Fail by increasing counter

**PING**  
You can diagnose network status by PING.

**PING**  
Show real time ping status.

# LCR – Black & White List

**Smart Web Manager**  
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**GSM LCR / Black List & White List**

**BlackList**

| Index | DialPattern | Control                  |
|-------|-------------|--------------------------|
| 0     | 888T        | <input type="checkbox"/> |
|       |             | Delete                   |
| 0     |             | Add                      |

**WhiteList**

| Index | DialPattern | Control                  |
|-------|-------------|--------------------------|
| 0     | 2...        | <input type="checkbox"/> |
|       |             | Delete                   |
| 0     |             | Apply                    |

**Information**

AddPac Technology  
Model : GS1002\_G2  
HW Version : 2.0  
SW 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 black & white list

**LCR Black & White List**  
Black List : The patterns are disallowed GSM outbound call.  
White List : The patterns are allowed GSM outbound call.

# LCR – Time Interval

The screenshot displays the Smart Web Manager interface for configuring GSM LCR / Time Interval Groups. The main content area is titled "GSM LCR / Time Interval Group" and contains a "TimeInterval" section. This section features a table with the following data:

| Group | Days     | StartTime(hh:mm) | EndTime(hh:mm) | Control                  |
|-------|----------|------------------|----------------|--------------------------|
| 0     | Weekdays | 00:00            | 23:59          | <input type="checkbox"/> |

Below the table, there are input fields for adding a new group: "0" for the group ID, "weekend" for the days, and "0" for both start and end times. There are also "Delete" and "Add" buttons.

A yellow callout box points to the table with the text: "Time Interval GSM outbound call is restricted by Time Interval".

The left sidebar contains a navigation menu with categories: System, Basic, Advanced, and Miscellaneous. The right sidebar contains an "Information" section with system details and a "Description" section with the text "Configure time interval group".

# LCR – Tariff Group

**Smart Web Manager**  
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**GSM LCR / Tariff Group**

**Tariff Group**

| Group | Time Group | Restore Call Limit |             | Accounting Period |           | Free Quota |     | Control                  |
|-------|------------|--------------------|-------------|-------------------|-----------|------------|-----|--------------------------|
| Type  | RestoreDay | First(sec)         | Others(sec) | Voice(min)        | SMS(E.A.) |            |     |                          |
| 0     | 0          | monthly            | 15          | 30                | 10        | 600        | 300 | <input type="checkbox"/> |
| 0     | 0          | daily              | 1           |                   |           |            |     | Delete<br>Add            |

**TariffPort**

| Port | TariffGroup |
|------|-------------|
| P0:0 | 0           |
| P0:1 | N.A.        |
| P0:2 | N.A.        |
| P0:3 | N.A.        |

**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 tariff group

**Time Interval Group**  
Time interval Group is adapted to this tariff group.

**Restore call limit**  
Quota restore time

**Accounting Period**  
Used to voice call.  
(ex)  
- configured as First (30 seconds ) Others(10 seconds )  
  
first connect -- 30 seconds are accounted.  
after 30 seconds -- 10 seconds are additionally accounted.  
after 10 seconds – 10 seconds are additionally accounted.  
and so on.

**Tariff Group**  
Tariff group is adapted to specific GSM port.

**Free quota**  
Free quota information.  
Current Usage information is supported at GSM Status.

# LCR – LCR Test

The screenshot shows the 'Smart Web Manager' interface for an LCR Test. The main area is titled 'LCR Test' and contains input fields for 'Caller:' (8888) and 'Called Number:' (9999), along with a 'Start' button. Below these is a log window showing the simulation results. Two yellow callout boxes provide additional context: one points to the 'Start' button and the log window, stating 'LCR Test LCR simulator', and another points to the log window, stating 'LCR Test Show real time simulation status.' The left sidebar shows a 'System' menu with options like Network Setup, Language, NAT, PPTP, and NTP, and an 'Advanced' menu with Gain & CID and GSM PIN. The right sidebar shows '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)) and a 'Description' section with 'LCR Test'.

**System**

- Network Setup
- Language
- NAT
- PPTP
- NTP

**LCR Test**

Caller:

Called Number:

```
< 1> LCR : =====
< 2> LCR : == GSM LCR(Least Cost Route) Simulator Start ==
< 3> LCR : =====
< 4> LCR : -- src digits : 8888(GSM) -> dst digits : 9999(GSM)
< 5> LCR : -- MatchAllProcess After Sorted
< 6> LCR : <0> id(3048) dest(T) prefer(0) selected(0)
< 7> LCR : -- Trying : <0> id(3048) dest(T)
< 8> LCR : -- Error: Outbound White Group(id:1) UnMatched
< 9> LCR : -----
< 10> LCR : -- Result : Fail
< 11> LCR : =====
< 12> LCR : == GSM LCR(Least Cost Route) Simulator End ==
< 13> LCR : =====
```

**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**

LCR Test

**LCR Test**  
LCR simulator

**LCR Test**  
Show real time simulation status.

**Advanced**

- Gain & CID
- GSM PIN

# SMS – Inbox

**Smart Web Manager**  
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**GSM SMS / InBox**

number of messages are 0 P0:0 OK

| Index | Sender | Received | Message | Select |
|-------|--------|----------|---------|--------|
|       |        |          |         | Delete |

**GSM SMS / In Box**

- total message
- unread messages (Blue color)
- received time
- content

**Information**

AddPac Technology  
Model : GS1002\_G2  
HW Version : 2.0  
SW 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**

# SMS – SMS New Message

The screenshot shows the 'Smart Web Manager' interface for sending a new SMS message. The main content area is titled 'GSM SMS / New Message' and contains a form with the following fields:

- Phone Number:
- Message:
- Port:

A red dashed box highlights the form fields. Below the form is a 'Send' button. A yellow callout box with a black border contains the text: 'New Message send a new message to the other GSM mobile phone.' An arrow points from this callout box to the form area.

The left sidebar contains a navigation menu with the following sections:

- System
  - Network Setup
  - Language
  - NAT
  - PPTP
  - NTP
- Basic
  - Protocol
  - Server SIP
  - SIP Registration
  - FXS Extension
  - GSM Extension
  - DTMF/CODEC
  - VoIP Dial Plan
  - GSM Dial Plan
  - Static Route
  - Hot Line
- 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

The right sidebar contains an 'Information' section with the following details:

- AddPac Technology
- Model: GS1002\_G2
- HW Version: 2.0
- SW 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)

Below the information section is a 'Description' section with a text area.



# Thank you!

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