VoiceFinder GSM Gateway

[CLI Command Manual]

Oct, 2010



AddPac Technology

www.addpac.com



AddPac GS3000 GSM Gateway

Note.

Product information can be changed at any time without notice. The information in this guide is provided as it is. It does not assure any kind of guarantee implicitly and explicitly under all circumstances. Regard of this solution guide, AddPac Technology and AddPac Technology solution supplier never guarantees the marketability and suitability of particular use, not invasive including; trade, usage, implicitly and explicitly. Yet, the feature on technological standard, and the information for the product guarantee and supply needs further consultation.



[Contents]

Chapter 1.	Interface Connection	6
Async Serial Interfac Ethernet Interface Co	e Connection	6 7
Chapter 2.	Proper Operation check .	8
Booting Process and Use Console Termino APOS Command Uso Administrator Mode Basic Setup APOS Upgrade	Operation Principle al by Using HyperTerminal age Command	
Chapter 3.	Console Command	23
Basic Command for Command for VoIP F Command for VoIP F GSM Command [Ca GSM Command [Bla GSM Command [LC GSM Command [LC GSM Command [LC	Network Setup Protocol [H.323] Protocol [SIP] Protocol [Monitoring] II] II] ck / White List] IIback Service] R(Least Cost Routing)] est]	23 25 31 35 37 41 43 43 46 48 51
Chapter 4.	Appendix	53
Console Port Signal	and Pin Out	53

[Chart]

<chart 2-1=""> Administrator Mode Command</chart>	19
<chart 4-1=""> Console Port Pin Out</chart>	53
<chart 4-2=""> Series Ethernet Cable Signal and Pin Out</chart>	54



[Picture Contents]

<picture 1-1=""> GS3000 Async</picture>	Serial Interface Connection	. 6
<picture 1-2=""> GS3000 LAN0/</picture>	LAN1 Interface connection	.7
<picture 2-1=""> MS-Windows T</picture>	erminal Emulator HyperTerminal	14
<picture 2-2=""> Enter Connec</picture>	ting Name in HyperTerminal	15
<picture 2-3=""> Setup Value w</picture>	when connecting console cable to serial port	16
<picture 2-4=""> COM1 Port Se</picture>	tup Example	17
<picture 2-5=""> APOS Image F</picture>	File Upgrade by Using FTP	21
<picture 4-1=""> 100Base-TX RJ</picture>	-45 Connector	54
<picture 4-2=""> Digital E1/T1 R</picture>	J45 Pin Out	55



Chapter 1. Interface Connection

Async Serial Interface Connection

Connect RS-232C Serial Console cable RJ-45 standard connector to console port. Connect serial port such as GS3000 control PC to opposite side serial connector.



<Picture 1-1> GS3000 Async Serial Interface Connection

Ethernet Interface Connection

For internet connection WAN interface, connect with WAN device (router or ADSL/cable modem) LAN interface and RJ45 standard UTP cable. Cross over may be used when connecting to the router or modem directly. Use direct-through cable when connecting to HUB.

Connect LAN0/LAN1 fast Ethernet interface by using RJ-45 connector.



<Picture 1-2> G\$3000 LAN0/LAN1 Interface connection



Chapter 2. Proper Operation check

Booting Process and Operation Principle

Following step shows booting process of GS3000.

- Gateway checks the CPU, memory, and interface.
- Boot Loader will be executed and find appropriate gateway s/w image file. Gateway is designed to upload s/w in default configuration.
- If s/w image file cannot be found in flash memory, boot loader will wait in boot mode until it downloads the proper gateway s/w. (Use TFTP or FTP protocol to download proper Next700 s/w)
- After gateway loading, gateway will be operated in accordance with saved information. Gateway will run with initial value if there is no saved setup information. For normal network operation, administrator must setup pertinent details.

After GS3000 installation and interface connection, power must be supplied. Make sure to connect power cable with GS3000. Do not connect GS3000 after providing power cable. Also, use 110V power cable if the power supply is 110V. GS3000 automatically recognizes both 110V and 220V so using proper power cable and additional operation is not necessary.



Following message shows after normal booting.

(Example) AP-GS3000 Booting Process

System Boot Loader, Version 5.2.9 Copyright (c) by AddPac Technology Co., Ltd. Since 1999.

[DUAL-BOOT] Start application (0xf1000100)...

System Bootstrap, Version 1.2

Decompressing the image:

VoiceFinder Router Series (GS3000_G2) Serial Number: GS3000_G2-fffe30 32BIT RISC Processor With 333MHz Clock 128 Mbytes System Memory 512 Kbytes System Boot Flash Memory 16 Mbytes System Flash Memory

1 RS232 Serial Console Interface

System main board ID is 0x0

GS3000_G2 System software Revision 8.41.03T Released at Fri Jul 23 10:11:57 2010 Program is 3497304 bytes, checksum is 0x1bdc47c6

Local Time : Wed Jul 28 11:48:19 2010 Copyright (c) by AddPac Technology Co., Ltd. Since 1999.



- SLOT[0] DETECT, ID(0x4)
- SLOT[1] DETECT, ID(0x4)
- SLOT[2] DETECT, ID(0x4)
- SLOT[3] DETECT, ID(0x4)
- SLOT[4] DETECT, ID(0x4)
- SLOT[5] DETECT, ID(0x4)
- SLOT[6] DETECT, ID(0x4)
- SLOT[7] DETECT, ID(0x4)
- SLOT[8] DETECT, ID(0x4)
- Allocating system mbuffer counter: 6144
- Kernel callwheelmask 0x3ff callwheelsize 1024
- Loading file system(ver2.2), flash-base: 0xffff0000 ram-base: 0x038499bc
- System utilization reference (28/28/28)
- Start Target Debug Server
- Attach FastEthernet Interface at Slot 0, Port 0-1, <0-0>/<0-1>
- FastEthernet0/0: link is up 100 Mbps (full duplex)
- Interface FastEthernet0/0, changed state to UP
- Interface FastEthernet0/1, changed state to DOWN
- Hardware Revision ID = 0
- Line Card equip status = 0xe00
- Slot (0) cardId=0 subId=4
- Slot (1) cardId=0 subId=4
- Slot (2) cardId=0 subId=4
- Slot (3) cardId=0 subId=4
- Slot (4) cardId=0 subId=4
- Slot (5) cardId=0 subId=4
- Slot (6) cardId=0 subId=4
- Slot (7) cardId=0 subId=4
- Slot (8) cardId=0 subId=4
- Slot (0) AP-N1-GSM4 : VoIP module 4 GSM (New 1)
- Slot (1) AP-N1-GSM4 : VoIP module 4 GSM (New 1)
- Slot (2) AP-N1-GSM4 : VoIP module 4 GSM (New 1)
- Slot (3) AP-N1-GSM4 : VoIP module 4 GSM (New 1)
- Slot (4) AP-N1-GSM4 : VoIP module 4 GSM (New 1)
- Slot (5) AP-N1-GSM4 : VoIP module 4 GSM (New 1)
- Slot (6) AP-N1-GSM4 : VoIP module 4 GSM (New 1)
- Slot (7) AP-N1-GSM4 : VoIP module 4 GSM (New 1)
- Slot (8) AP-N1-GSM4 : VoIP module 4 GSM (New 1)



VOIP_INTERFACE_UP : (172.17.114.130)		
Not Available: status(1) interface(0)		
VOIP_INTERFACE_DOWN		
init0		
Wait	init0	Wait
Start File Transfer Protocol Server (listen tcp/21)		
GsSim Start		
RTA_Daemon start		
Slot (8) Module type : GSM		
Slot (7) Module type : GSM		
Slot (6) Module type : GSM		
Slot (5) Module type : GSM		
Slot (4) Module type : GSM		
Slot (3) Module type : GSM		
Slot (2) Module type : GSM		
Slot (1) Module type : GSM		
Slot (0) Module type : GSM		

Not Available: status(1) interface(0)

The port is not available The port is not available The port is not available The port is not available

GsSim Ready

Press RETURN to get started.

RTA Module Ready DSP S/W download (0): OK DSP S/W download (1): OK DSP S/W download (2): OK DSP S/W download (3): OK DSP S/W download (4): OK DSP S/W download (5): OK DSP S/W download (6): OK



- DSP S/W download (8): OK
- QS (0) install ... OK QS (1) install ... OK QS (2) install ... OK QS (3) install ... OK QS (4) install ... OK QS (5) install ... OK QS (6) install ... OK QS (7) install ... OK

Gatekeeper shutdowned.

VoIP in service.

GSM-0/0: MODULE ID =	MULTIBAND	900E	1800
GSM-4/2: MODULE ID =	MULTIBAND	900E	1800
GSM-7/3: MODULE ID =	MULTIBAND	900E	1800
GSM-1/0: MODULE ID =	MULTIBAND	900E	1800
GSM-6/2: MODULE ID =	MULTIBAND	900E	1800
GSM-8/2: MODULE ID =	MULTIBAND	900E	1800
GSM-2/3: MODULE ID =	MULTIBAND	G850	1900
GSM-3/0: MODULE ID =	MULTIBAND	900E	1800
GSM-5/2: MODULE ID =	MULTIBAND	900E	1800
GSM-6/3: MODULE ID =	MULTIBAND	G850	1900
GSM-5/3: MODULE ID =	MULTIBAND	G850	1900
GSM-0/1: MODULE ID =	MULTIBAND	G850	1900
GSM-3/3: MODULE ID =	MULTIBAND	900E	1800
GSM-4/3: MODULE ID =	MULTIBAND	900E	1800
GSM-6/0: MODULE ID =	MULTIBAND	900E	1800
GSM-8/0: MODULE ID =	MULTIBAND	900E	1800
GSM-1/1: MODULE ID =	MULTIBAND	900E	1800
GSM-7/0: MODULE ID =	MULTIBAND	900E	1800
GSM-8/3: MODULE ID =	MULTIBAND	900E	1800
GSM-1/3: MODULE ID =	MULTIBAND	900E	1800
GSM-7/1: MODULE ID =	MULTIBAND	900E	1800
GSM-8/1: MODULE ID =	MULTIBAND	G850	1900
GSM-7/2: MODULE ID =	MULTIBAND	900E	1800
GSM-4/1: MODULE ID =	MULTIBAND	900E	1800
GSM-1/2: MODULE ID =	MULTIBAND	900E	1800



GSM-2/0: MODULE ID =	MULTIBAND	900E	1800					
GSM-4/0: MODULE ID =	MULTIBAND	900E	1800					
GSM-3/1: MODULE ID =	GSM-3/1: MODULE ID = MULTIBAND 900E 1800							
GSM-3/2: MODULE ID =	MULTIBAND	900E	1800					
GSM-2/1: MODULE ID =	MULTIBAND	900E	1800					
GSM-0/2: MODULE ID =	MULTIBAND	G850	1900					
GSM-2/2: MODULE ID =	MULTIBAND	900E	1800					
GSM-0/3: MODULE ID =	MULTIBAND	900E	1800					
GSM-6/1: MODULE ID =	MULTIBAND	900E	1800					
GSM-5/1: MODULE ID =	GSM-5/1: MODULE ID = MULTIBAND 900E 1800							
GSM-5/0: MODULE ID = MULTIBAND 900E 1800								
GSM-0/0: SIM READY								
GSM-0/1: SIM READY								
GSM-0/2: SIM READY								
GSM-0/3: SIM READY								
Interface GSM-0/0, changed state to UP								
Interface GSM-0/1, changed state to UP								
Interface GSM-0/2, changed state to UP								
Interface GSM-0/3, changed state to UP								
Welcome, APOS(tm) Kernel Version 8.41.03T.								

```
Copyright (c) 1999-2010 AddPac Technology Co., Ltd.
```

Login:

Password:

When log-in message shows at the end, enter basic log-in value "<u>root</u>" and basic password value "<u>router</u>" then log-in is completed. After log-in procedure, prompt "**Router**>" is displayed in console terminal.

Prompt which displays on VoiceFInder GSM Gateway includes: "router>" and "router#". If prompt is displayed as ">", logged-in user has other authority other than "admin". A user cannot use command which changes gateway setup and a user only has minimum authority. If the prompt is displayed as "#", current logged-in user has "admin" authority and uses all gateway functions.

If you log in as Admin, Admin cannot change all gateway setup. So, we recommend to change admin account password basic value for security.



Use Console Terminal by Using HyperTerminal

Terminal Emulator Application must be setup if PC is being used as console terminal. It uses Hyper Terminal Application when using MS-Windows.



<Picture 2-1> MS-Windows Terminal Emulator HyperTerminal



Setup a new connection name after running HyperTerminal. Setup name can be decided as you like. We will decide as AddPac.

New Connectio	n - HyperTerminal all Transfer Help		-D×
06 93	<u>0</u> 8		
		Image: New Connection Image: New Connection Name: AddPac Icon: Image: New Connection Icon:	
Disconnected	Auto detect	Auto detect SCROLL CAPS NUM Capture Print echo	

<Picture 2-2> Enter Connecting Name in HyperTerminal

Now, we setup interface which is connected to console cable.

Console Cable is usually connected to PC RS-232C 9Pin Serial Port. Choose a right port in accordance with user environment. We will connect to COM1.

AddPac - HyperTerr	ninal Transfer Help	Connect To AddPa Enter details for Gountry/region Arga code: Phone numbe Cognect using	ic or the phone r it Korea (Re 02 c c c c c c c out	number that public of)	at you wa 192)	nt to dial:			- D ×
Sconnected	Auto detect	Auto detect	SCROLL	CAPS	NUM	Capture	Print echo	-1	

<Picture 2-3> Setup Value when connecting console cable to serial port

We setup each setup values in interface registration information. We make COM1 as standard.

		rany	siop blis	Flow Control
9600	8	None	1	None
AddPac - HyperTerminal File Edit View Call Transfer H □ : ((((((((((((((((((((((((((((((((((((elp			
	COMI Properties Port Settings Bits per sec Data Pr Stop Flow cor	ond: 9600 bits: 8 arity: None bits: 1 atrot: None Rest OK Cancel	?×	

<Picture 2-4> COM1 Port Setup Example

Booting message shows on Hyper Terminal after setup.



APOS Command Usage

NOTE AddPac technology devices are all equipped with APOS (AddPac Operating System). So basic CLI (Command Line Interface) environments are all identical.

All command in gateway can be connected to console or telnet terminal (VT-100 terminal). Command provides to view system restriction items, user mode to provide access function, look at the system status. Administrator mode to use system debuggin function and change the setup environment or setup a new environment.

Following characteristics are gateway related command input.

- It automatically recognizes without typing all command. For example, to run "show" command, type "sh" or "sho". It will automatically recognizes as "show"
- It provides on-line help function so that when user enters incorrect system command, possible items and commands will be shown.
- More function provides to display unable information on screen.
- It provides Help and "?" function to display all possible command and explanation.
- It provides "History" function. User may use Prompt number without retyping command.
- System command structures are divided into 3 modes. Each mode has different command. Command for each mode is as below.



Administrator Mode Command

Administrator mode command can only be used for logged-in administrator. To access system setup mode under looged-in administrator mode. In this mode, it shows more information in accordance with options.

You may use all command in administrator mode.

Administrator mode prompt is displayed as "GS3000#".

Command	Explanation
clear	Command to initialize interface counter, statistics
configure	Enter as setup mode
сору	Copy running config as startup config
debug	Overall system debug Command
disconnect	Command to close VTY connection
end	Enter as administrator mode
erase	Delete config file
exit	Move to previous step
ftp	Access ftp client
help	View APOS Help
no	Command to delete current setup
ping	Network connection confirmation tool
reboot	System rebooting command
show	Command to view the system operation status and setup
SHOW	status
telnet	telnet access command
tftp	Command to send file to tftp server
traceroute	Command to check IPv4 routing route
who	Command display accessed user with vty
write	Command to save operating configuration
undebug	Command to deactivate the activated debug
User	Command to add/modify/delete system user

<Chart 2-1> Administrator Mode Command



Basic Setup

Host Name Setup

You may change prompt name in CLI environment when you access into console or telnet. Host name considers as very important when you manage several devices with telnet access.

Router# configure terminal Router(config)# Router(config)# hostname {name} Router(config)#

Clock Setup

Setup system time. You should set it correctly because this time will be displayed on system operation time, debug and log.

Router# configure terminal Router(config)# Router(config)# clock {Year} {Month} {Day} {Hour} {Minute} {Second} Router(config)#

• User Management

User management can be used when accessing telnet, FTP, and SAMBA. You may have to use user password/ID which is known to administrator and if it is exposed to others, device would not operate as it is.

Router# configure terminal

Router(config)#

Router(config)# username {add} <login-name> <password> {admin|high|normal|low} Router(config)#



APOS Upgrade

AddPac Gateway products allow FTP method approach for APOS image file transmit. Also, a relevant protocol can independently on/off the service.

AddPac GSM Gateway upload/download network configuration is as below.



<Picture 2-5> APOS Image File Upgrade by Using FTP

• FTP Service Activation

Activate FTP service first to execute the APOS upload/download.

Router(config)# Router(config)# ftp server	Router# configure terminal
Router(config)# ftp server	Router(config)#
D = (x + y) (x - y) (x + y)	Router(config)# ftp server
Kouter(coning)#	Router(config)#



• APOS Upload

Upload gateway APOS in PC with ftp.

D:₩ >dir			
2007-11-15	05:21p	<dir> .</dir>	
2007-11-15	05:21p	<dir></dir>	
2007-11-15	05:21p	3,098,820	GS3000_v8_51_001.bin
D.#\>			
D:₩> ftn 172	17 201 115		
Connected to	$172\ 17\ 201^{-1}$	115	
220 GS3000 F	FTP server (Version 8.51) readv	
User (172.17.	.201.115:(nor	ne)): root	
331 Password	d required for	root.	
Password:	-		
230 User root	t logged in.		
ftp> binary			
200 Type set	to I.		
ftp>			
ftp> put GS30	000_g2_v8_51	_001.bin	
200 PORT co	mmand succe	essful.	
150 Opening I	BINARY mod	e data connection fo	or 'GS3000_g2_v8_51_001.bin '.
226 Transfer	complete.		
ftp: 3098820	bytes sent in	2.47Seconds 1039.	.51Kbytes/sec.
ftp> quit			
221 Goodbye.			
D:₩>			



Chapter 3. Console Command

Basic Command for Network Setup

• Command to confirm after IP, default route setup

Following configuration shows network setup for basic communication which identifies the proper network connectivity with default gateway ping test.

Router# configure terminal Router(config)# interface FastEthernet 0/0 Router(config-if)# ip address 172.17.201.115 255.255.0.0 Router(config-if)# **exit** Router(config)# ip route 0.0.0.0 0.0.0.0 172.17.1.1 Router(config)# end Router# Router# Router# write Do you want to WRITE configuration ? [y|n] yWriting configuration....done Router# Router# ping 172.17.1.1 PING 172.17.1.1 (172.17.1.1): 56 data bytes 64 bytes from 172.17.1.1: icmp_seq=1 ttl=255 time=2 ms 64 bytes from 172.17.1.1: icmp_seq=2 ttl=255 time=1 ms 64 bytes from 172.17.1.1: icmp_seq=3 ttl=255 time=1 ms 64 bytes from 172.17.1.1: icmp_seq=4 ttl=255 time=1 ms 64 bytes from 172.17.1.1: icmp_seq=5 ttl=255 time=1 ms --- 172.17.1.1 ping statistics ---5 packets transmitted, 5 packets received, 0% packet loss round-trip min/avg/max = 1.73/2.027/2.75 ms Router#



• PPPoE Configuration

Step	Command	Explanation
1	# configure terminal (config)#	Change to APOS command input mode.
2	(config)# interface fastethernet 0/0 (config-if)#	Start fast Ethernet 0/0 setup.
3	(config-if)# no ip address	No IP address setup.
4	(config-if)# pppoe enable	Activate network protocol PPPoEdmf.
5	(config-if)# encapsulation ppp	Assign network protocol as PPP.
6	(config-if)# ppp authentication pap callin	Setup PPP certification method as PAP.
7	(config-if)# ppp ipcp default-router	Setup to receive default router IP address from PPP server.
8	(config-if)# ppp ipcp dns request	Setup to receive DNS IP address from PPP server.
9	(config-if)# ppp pap sent-username addpac password 1234	Setup PAP user ID as "addpac", password as"1234".

• DHCP Configuration

Step	Command	Explanation
1	# configure terminal Enter configuration commands, one per line. End with CNTL/Z (config)#	Change to APOS command input mode.
2	(config)# interface fastethernet 0/0 (config-if)#	Start setting up fast Ethernet interface 0/0.
3	(config-if)# ip address dhcp (config-if)#	Setup to assign IP address through DHCP client setup.

• Static IP Configuration

Step	Command	Explanation
1	# configure terminal Enter configuration commands, one per line. End with CNTL/Z (config)#	Change to APOS command input mode.
2	(config)# interface fastethernet 0/0 (config-if)#	Start setting up fast Ethernet interface 0/0.
3	(config-if)# ip address 192.168.1.2 255.255.255.0	Setup IP address.
4	(config-if)# ip route 0.0.0.0 0.0.0.0 192.168.1.1	Setup default router.

Command for VoIP Protocol [H.323]





A necessary gateway setup parameter is as follow.

→ Gateway 의 IP Address / Dial-peer Voip / Dial-peer Pots / VoIP Interface

example configuration(HQ)	example configuration(Branch_A)	example configuration(Branch_B)
example configuration(HQ) hostname HQ Interface fastethernet 0/0 ip address 194.168.1.2 255.255.255.0 ip route 0.0.0.0 0.0.0 194.168.1.1 idial-peer voice 0 pots destination-pattern 8425683848 port 0/0 i dial-peer voice 1000 voip destination-pattern 8325683848 session target ip 193.168.1.2 dtmf-relay h245-alphanumeric i dial-peer voice 1001 volp destination-pattern 82T session target ip 192.168.1.2	example configuration(Branch_A) hostname BA interface fastethernet 0/0 Ip address 193.168.1.2 255.255.255.0 ip route 0.0.0.0 0.0.0 193.168.1.1 dial-peer voice 0 pots destination-pattern 8325683848 port 1/0 I dial-peer voice 1000 voip destination-pattern 82 session target ip 192.168.1.2 dtmf-relay h245-alphanumeric I dial-peer voice 1001 voip destination-pattern 8425683848	example configuration(Branch_B) hostname BB interface ether0.0 ip address 192.168.1.2 255.255.255.0 ip route 0.0.0 0 0.0.0 192.168.1.1 dial-peer voice 0 pots destination-pattern 8225683848 port 0/0 I dial-peer voice 1000 voip destination-pattern 8425683848 session target ip 194.168.1.2 dtmf-relay h245-alphanumeric I dial-peer voice 1001 voip destination-pattern 8325683848
session target ip 192.168.1.2 dtmf-relay h245-alphanumeric I voip-interface ip fastethernet 0/0 !	8425683848 session target ip 194.168.1.2 dtmf-relay h245-alphanumeric I volp-interface ip fastethernet 0/0 '	8325683848 session target ip 193.168.1.2 dtmf-relay h245-alphanumeric ! voip-interface ip fastethernet 0/0 !
	-	· · ·

Step	Command	Explanation
1	HQ(config)# dial-peer voice 0 pots HQ(config-dialpeer-pots-0)#	Assign serial number 0 to physical voice interface to set one phone line. (Serial number can be assigned from 0 to 65,535, normally assign from 0 when setting voice interface)
2	HQ(config-dialpeer-pots-0)# destination- pattern 8425683848	Grant phone number to assigned serial number 0.
3	HQ(config-dialpeer-pots-0)# port 0/0	Setup VoIP physical voice port which responds to assigned serial number 0. (voice port types are may vary depending on devices)
4	HQ(config-dialpeer-pots-0)# dial-peer voice 1000 voip HO(config-dialpeer-voip-1000)#	To setup one VoIP call, setup serial number 1000 to VoIP interface serial number. (Serial number can be assigned from 0 to 65535. Assign from 1000 when setting VoIP interface.)
5	HQ(config-dialpeer-voip-1000)# destination- pattern 8325683848	Grant phone number 8325683848 to assigned serial number 1000.
6	HQ(config-dialpeer-voip-1000)# session target ip 193.168.1.2	Send VoIP call setup to assigned serial number 1000 with gatekeeper.
7	HQ(config-dialpeer-voip-1000)# dtmf-relay h245-alphanumeric	Setups DTMF transmit method as H.245 Alphanumeric to assigned serial number 1000.



8	HQ(config-dialpeer-voip-1000)# dial-peer voice 1001 voip	To setup other VoIP call, assign serial number 1001 to VoIP interface.
9	HQ(config-dialpeer-voip-1001)# destination- pattern 82T	Setup if the phone number starts with 82 to assigned serial number 1001.
10	HQ(config-dialpeer-voip-1001)# session target ip 192.168.1.2	Send VoIP call setup with 192.168.1.2 to assigned serial number 1001.
11	HQ(config-dialpeer-voip-1001)# dtmf-relay h245-alphanumeric	Setups DTMF transmit method as H.245 Alphanumeric to assigned serial number 1001 (default).
12	HQ(config)# voip-interface ip fastethernet 0/0	Setup VoIP interface as fast Ethernet port. (Default).
13	HQ(config)# ip route 0.0.0.0 0.0.0.0 192.168.1.1 HQ#	Setup default router as 192.168.1.1.
14	HQ(config)# exit HQ#	Exit from APOS command input mode.









Configuration checking	Configuration checking
HO# show gateway	number of pots peers = 1
Gatekeeper Registration Information	number of volp peers = 2
H.323 id = addpac	number of volp cers = 2
gatekeeper Registration option = enabled	number of codec classes = 0
gatekeeper registration status :	number of alternate gatekeepers = 1
registered.	number of alternate gatekeepers = 1
last registration reject information from gatekeeper	number of alternate gatekeepers = 1
ConfigAsHOREgistration (Aug 9 03:02:43)	number of alternate gatekeepers = 1
Gatekeeper list :	number of current calls = 0
199.168.1.1 1719 priority(128) by user	Announcement Option
Local allases	language = korean
[1] H323D + addpac-HQ	element : delyed del = disabled
[2] 8425693848	element : connection fail = disabled
Technical prefixes	element : connection fail = disabled
Gateway Information	element : connection fail = disabled
status = Int 1 (waiting for setting IP address on a VoIP Interface)	thit (Initial dight timer) = 10 sec.
product name = AddPac VOIP	tring (ring timer) = 30 sec.
product version = 6.12	tring (ring timer) = 30 sec.
endpoint type = gateway	tras (RAS msg ack timer) = 6 sec.
discovery (send GRQ) = disabled	titti (RAS Time To Live timer) = 60 sec.
ARQ option = no inq	treg2 (GK Registration retry timer) = 20 sec.
lightweight IRR = disabled	treg2 (GK Registration retry timer) = 20 sec.
TTL margin = 20 %	treg2 (GK Registration retry timer) = 20 sec.
h323 call start mode = fast	treg2 (GK Registration retry timer) = 120 sec.
h323 call start mode = fast	treg2 (GK Registration retry timer) = 120 sec.
h323 call channelim gmode = enabled	treg2 (GK Registration retry timer) = 120 sec.
h232 call channelim gmode = late	treg1 (polling timer on trunk or polling type connection) = 130 sec.
h323 response msg = default	dtmf duration = 150 msec.
system fax mate = 138	cras (RAS retry counter) = 3
system fax rate (bps) = 9600	Remote Call Log (syslog)
system T.38 fax redundancy = 0	primary server =
force to send stat1245 = enabled	interval = 0 minutes
dialPeer hunt algorithm = longest - preference - random	cdr format type = 0
translate voip incoming calling number = -1	Assigned VoIP TCP/UDP ports
local ringback tone = normal	minimized assign = no
end of digit = #	Q.9331 signalling port (TCP src) = 10000 - 10999
ip address prefix = #	Q.9331 signalling port (TCP src) = 10000 - 14999
voice confirmed connect on FXO/E&M = disabled	RAS port and IRR port (UDP listen) = 23000, 22001
number of ports = 1	RTP/RTCP port (UDP listen) = 23000 - 24999

Step	Command	Explanation
1	HQ(config)# dial-peer voice 0 pots	Assign serial number 0 to physical voice interface for setting up one phone line. (serial number can be assigned from 0 to 65,535. For voice interface setup, it usually starts from 0)
2	HQ(config-dialpeer-pots-0)# destination- pattern 8225683848	Grant phone number to assigned serial number 0.
3	HQ(config-dialpeer-pots-0)# port 0/0	Set VoIP device physical voice port which responds to assigned serial number 0. (Voice port types/numbers are may vary.)
4	HQ(config-dialpeer-pots-0)# dial-peer voice 1000 voip	Set serial number 1000 to VoIP interface for VoIP call setup. (serial number can be assigned from 0 to 65535. Assign from 1000 for VoIP interface setup.)
5	HQ(config-dialpeer-voip-1000)# destination- pattern 8325683848	Grant phone number 8325683848 to assigned serial number 1000.
6	HQ(config-dialpeer-voip-1000)# session target ras	Send VoIP call setup to assigned serial number 1000 with gatekeeper.
7	HQ(config-dialpeer-voip-1000)# dtmf-relay h245-alphanumeric	Setups DTMF transmit method as H.245 Alphanumeric to assigned serial number 1000.



8	HQ(config-dialpeer-voip-1000)# dial-peer voice 1001 voip	Assign serial number 1001 to VoIP interface for VoIP call setup.
9	HQ(config-dialpeer-voip-1001)# destination- pattern 82T	Setup it the phone number starts with 82 to assigned serial number 1001.
10	HQ(config-dialpeer-voip-1001)# session target	Send VoIP call setup to assigned serial number 1001 with gatekeeper.
11	HQ(config-dialpeer-voip-1001)# dtmf-relay h245-alphanumeric	Setup DTMF transmission method as H.245 Alphanumeric to assigned serial number 1001.
12	HQ(config-dialpeer-voip-1001)# gateway	Change to external gatekeeper setup mode.
13	HQ(config-gateway)# h323-id addpac-BB	Setup H.323 ID.
14	HQ(config-gateway)# gkip 199.168.1.1 1719 128	Setup external gatekeeper IP address.
15	HQ(config-gateway)# register	Register in external gatekeeper.
16	HQ(config-gateway)# voip-interface ip fastethernet 0/0	Set VoIP interface as Ethernet 0.0 port.
17	HQ(config)# exit HQ#	Exit from APOS command input mode.



Command for VoIP Protocol [SIP]

• SIP – Peer to Peer



View SIP Related Command

Step	Command		Explanation
	# config (config)#		Enter as APOS command setup mode.
1			
	(config)# sip-ua		Enter as SIP user agent command setup mode
2	(config-sip-ua)# ?		and confirm command with "?"
	no set to default configuration	Delete command which is already setup in SIP	
		user agent.	
	register	try registration to sip registrar	Try to register in SIP-Server.



signaling-port	set SIP signaling port (default 5060)	Change SIP signaling port. (Default: 5060)
sip-server	Configure a SIP Server Interface	Enter SIP Server IP address or DNS server.
sip-username	Set Username of SIP User Agent	Register SIP user name.
sip-password	Set Password of SIP User Agent	Register SIP password.
timeout	Set timeout value	Setup register retry timeout value.
end	Go to Top menu	Return to administrator mode.
exit	Exit from the EXEC	Return to APOS command setup mode.

Step	Command	Explanation
1	HQ(config-dialpeer-pots-0)# dial-peer voice 1000 voip	Assign serial number 1000 to VOIP interface for VoIP call setup. (serial number can be assigned from 0 to 65535. Assign from 1000 for VoIP interface setup.)
2	HQ(config-dialpeer-voip-1000)# destination-pattern T	
3	HQ(config-dialpeer-voip-1000)# session target ip 192.168.1.2	Setup VoIP destination address as 192.168.1.2 which responds to destination pattern.
4	HQ(config-dialpeer-voip-1000)# session protocol sip	Setup VoIP protocol as SIP which is relevant to destination pattern. If you setup as "no protocol sip", it will send to h.323 message.
5	HQ(config-dialpeer-voip-1000)# dtmf-relay rtp-2833	Send DTMF transmission method with RFC2833 RTP payload to assigned serial number 1000. (currently, SIP mode only offers 2833 for dtmf signal).

• SIP – SIP Server



Sample Configuration(HQ) about SIP Server	Sample Configuration(Branch_A) about SIP Server
hostname HO	hastname BA
interface fastethernet0/0 p address 194.168.1.2255.255.255.0	interface ⁻ cstethernetC/0 Ip acdress 192.168.1.2255.255.255.0
lp route 0.0.0.0 0.0.0.0 194.168.1.1	Ip route 0.0.0.00.0.0.0 192.168.1.1
dnshost nameserver 199.168.1.2	dnshost nameserver199.168.1.2
i I Pots peer configuration.	i I Pots peer configuration.
i dial-peer voice 0 pots destination-partern 8425683848 port 3/0 l	dial-peer voice 0 pots destination-pattern 8225683848 port 0/0 I
l Voip peer configuration.	l ! Voip peer conīguration.
dial-peer voice 1000 voip destinction-patiem T session target sip-server session protocol sip dimt-relay rip-2833	diai-peer voice 1000 voip destination-pattern T session target sip-server session protocol sip dimf-relay rip-2833
Gareway configuration.	I Gatewcy configuration.
ISPUA configuration.	I SIP LA configuration.
: slp-ua slp-usemame 8225683848 slp-password AddPac-HO slp-server proxy.addpac-tesLcom register e 164 ! voip-nterface ip fastethernet 0/0	sip-ua sip-usemame 6425663848 sip-password AddPac-A sip-server 192.168.1.1 register e164 ! vcip-interface ip fastethernet0/0

SIP Proxy Server Configuration

Step	Command	Explanation
1	HQ(config)# dnshost nameserver 199.168.1.2	Register DNS server IP.
2	HQ(config-dialpeer-pots-0)# dial-peer voice 1000 voip	Assign serial number 1000 to VoIP interface for VOIP call setup. (serial number can be assigned



		from 0 to 65535. Assign from 1000 for VoIP interface setup.)
3	HQ(config-dialpeer-voip-1000)# destination-pattern T	
4	HQ(config-dialpeer-voip-1000)# session target sip-server	Setup VoIP destination address as SIP server which responds to destination pattern.
5	HQ(config-dialpeer-voip-1000)# session protocol sip	Setup VoIP protocol as SIP which is relevant to destination pattern. If you setup "no protocol sip", it will send to h.323 message.
6	HQ(config-dialpeer-voip-1000)# dtmf-relay rtp-2833	Send DTMF transmission method with RFC 2833 RTP payload to assigned serial number 1000 (Currently, 2833 only provides dtmf signal in SIP mode).
7	HQ(config-dialpeer-voip-1000)# sip-ua	Enter as SIP UA setup mode
8	HQ(config-sip-ua)# sip-username 8425683848	Register SIP user name
9	HQ(config-sip-ua)# sip-password AddPac-HQ	Register password
10	HQ(config-sip-ua)# sip-server 192.168.1.1	Setup SIP server IP (DNS server must be setup if you set up as domain name)
11	HQ(config-sip-ua)# register e164	Register to SIP server as E.164.

Command for VoIP Protocol [Monitoring]

Step	Command	Explanation
1	Router# show voice port <slot> <port> / all / summary</port></slot>	Check voice port information
2	ROUTER# show run	Check all setup information
3	ROUTER# show sip	Check SIP register status and gateway value
4	ROUTER# show gateway	Check H.323 register status and gateway value
5	ROUTER# show version	Check gateway version
6	ROUTER# show clock	Check gateway current time/running time
7	ROUTER# show interface	Check gateway interface status
8	ROUTER# show dial-peer voice run / summary	Check Dial-peer setup information
9	ROUTER# show username	Check gateway account ID / PW

• VoIP - Monitoring command



	•	VoIP	- Debugging	command
--	---	------	-------------	---------

Step	Command	Explanation
1	ROUTER# debug voip call	Debug On for VoIP call
2	ROUTER# debug rta ipc	Debug On for port
3	ROUTER# debug voip sip	Debug On for SIP
4	ROUTER# debug voip h225-asn1	Debug On for H.225
5	ROUTER# debug voip h245-asn1	Debug On for H.245
5	ROUTER# terminal monitor	Activate monitoring for Debug Commands
5	ROUTER# no debug all	Debug Off



GSM Command [Call]



• GSM - FXS to GSM Call Outbound Call (1 Stage)

: Making call to mobile phone from analog phone connected to FXS directly.

Outbound Call (2 Stage)

: Making call to mobile phone from analog phone connected to FXS after hearing of 2nd dial tone from AP-GS1002

```
1 Stage Call Sample Configuration2 StagIhostname AP-GS1004Iinterface fastethemet 0/0interface fastethemet 0/2ip address 194.168.1.2 255.255.255.0IIinterface fastethemet 0/0ip route 0.0.0.0 0.0.0 194.168.1.1Iidial-peer voice 0 potsdial-peer voice 0 potsdestination-pattern 1000jort 0/0Idial-peer voice 1 potsdestination-pattern TPort 1/0Port 1/0I
```

```
2 Stage Call Sample Configuration
```



Step	Command	Explanation
1	AP-GS1004(config)# dial-peer voice 1 pots AP-GS1004(config-dialpeer-pots-1)#	Assign serial number 1 to physical voice interface for phone line setup (serial number can be assigned from 0 to 65,535. Assign from 0 for voice interface setup.)
2	AP-GS1004(config-dialpeer-pots-1)# destination-pattern T	Grant phone number to assigned serial number1.
3	AP-GS1004(config-dialpeer-pots-1)#Port 1/0	Set physical GSM port which responds to assigned serial number 1. (Voice port types/numbers may vary)
4	AP-GS1004(config)# exit AP-GS1004#	Exit from input mode.



• GSM - VoIP to GSM Call

1 Stage Call Sample Configuration (SIP)	1 Stage Call Sample Configuration (H.323)
!	
hostname AP-GS1004	hostname AP-GS1004
l	
interface fastethernet 0/0	interface fastethernet 0/0
ip address 194.168.1.2 255.255.255.0	ip address 194.168.1.2 255.255.255.0
l	
ip route 0.0.0.0 0.0.0.0 194.168.1.1	ip route 0.0.0.0 0.0.0.0 194.168.1.1
l	
dial-peer voice 0 pots	dial-peer voice 0 pots
destination-pattern T	destination-pattern T
Port 1/0	Port 1/0
!	!
Voip peer configuration.	Voip peer configuration.
	!
dial-page voice 1000 voip	dial-pager voice 1000 voip
destination-pattern 1000 session target ip 194.168.1.3 session protocol sip dtmf-relay rtp-2833	destination-pattern 1000 session target ip 194.168.1.3 dtmf-relay h245-alphanumeric

Step	Command	Explanation
1	AP-GS1004(config)# dial-peer voice 0 pots AP-GS1004(config-dialpeer-pots-0)#	Assign serial number 0 to physical voice interface for phone line setup. (serial number can be assigned from 0 to 65,535. Assign from 0 for voice interface setup.)
2	AP-GS1004(config-dialpeer-pots-0)# destination-pattern T	Grant phone number to assigned serial number 0.
3	AP-GS1004(config-dialpeer-pots-0)# port 1/0	Set physical GSM port to assigned serial number 0. (Voice port types/numbers are may vary)
5	AP-GS1004(config)# dial-peer voice 1000 voip AP-GS1004(config-dialpeer-voip-1000)#	Assign serial number 1000 to VoIP interface for VOIP call setup. (serial number can be assigned from 0 to 65535. Assign from 1000 for VOIP interface setup.
6	AP-GS1004(config-dialpeer-voip-1000)#destination-pattern T	It means all input numbers.
7	AP-GS1004(config-dialpeer-voip-1000)#session target ip 194.168.1.3	Setup VoIP destination address as 192.168.1.2 which responds to destination pattern.
8	AP-GS1004(config)# session protocol sip	Setup VoIP protocol as SIP which responds to destination pattern. If you setup as "no protocol sip", it will send to h.323 message.
9	AP-GS1004(config)# dtmf rtp-2833	Send DTMF transmission method as RFC 2833



		RTP payload to assigned serial number 1000 (Currently, 2833 only provides dtmf signal in SIP mode).
10	AP-GS1004(config)# exit AP-GS1004#	Exit from input mode



GSM Command [Black / White List]

• GSM - Black / White List



Step	Command	Explanation
1	AP-GS1004(config)# dialpattern-group 0 AP-GS1004(config-dialpattern-group#0)#	Create Black List Group.
2	AP-GS1004(config-dialpattern-group#0)# pattern 0 1000	Enter pattern number which is relevant to Black List.
3	AP-GS1004(config-dialpattern-group#0)# exit	Return to configuration mode.
4	AP-GS1004(config)# voice-port 1/0	Enter to applicable port.
5	AP-GS1004(config-voice-port-1/0)# gsm inbound black list-group 0	Apply regarding inbound call.
6	AP-GS1004(config-voice-port-1/0)# gsm outbound black list-group 0	Apply regarding outbound call.
7	AP-GS1004(config)# dialpattern-group 1 AP-GS1004(config-dialpattern-group#0)#	Create White List Group.
8	AP-GS1004(config-dialpattern-group#0)# pattern 0 2000	Enter patter number which is relevant to White List.
9	AP-GS1004(config)# voice-port 1/0	Enter to applicable port.
10	AP-GS1004(config-voice-port-1/0)# gsm inbound white	Apply regarding inbound call.



	list-group 1	
11	AP-GS1004(config-voice-port-1/0)# gsm outbound white list-group 1	Apply regarding outbound call.
12	AP-GS1004(config-voice-port-1/0)# exit AP-GS1004(config)#	Exit from input mode.



GSM Command [Callback Service]

```
• GSM - WEB Callback
```

```
WEB Callback Sample Configuration
!
hostname AP-GS1004
!
interface fastethernet 0/0
ip address 194.168.1.2 255.255.255.0
!
ip route 0.0.0.0 0.0.0.0 194.168.1.1
! Gateway configuration
!
gateway
 gsm-call-service-white-list-group 2
!
!
! Dial Pattern Group configuration.
!
dialpattern-group 2
                  1000
 pattern 0
!
```

Step	Command	Explanation
1	AP-GS1004(config)# dialpattern-group 2 AP-GS1004(config-dialpattern-group#0)#	Create WEB Callback Group.
2	AP-GS1004(config-dialpattern-group#0)# pattern 0 1000	Enter the number for WEB Callback
3	AP-GS1004(config-dialpattern-group#0)# exit	Return to configuration mode.
4	AP-GS1004(config)# gateway	Enter to gateway setup.
5	AP-GS1004(config-gateway)# gsm-call-service-white-list- group 2	Apply WEB Callback Group



6	AP-GS1004(config)# gateway AP-GS1004(config-gateway)# gsm-call-service 1000 2000	Execute WEB Callback.

• GSM – GSM Callback

```
GSM Callback Sample Configuration
!
hostname AP-GS1004
!
interface fastethernet 0/0
 ip address 194.168.1.2 255.255.255.0
!
ip route 0.0.0.0 0.0.0.0 194.168.1.1
! Voice port configuration
!
! GSM
Voice-port 1/0
  gsm-call-back-white-list-group 3
!
! Dial Pattern Group configuration.
!
Dialpattern-group 3
  pattern 0
                   2000
!
```

Step	Command	Explanation
1	AP-GS1004(config)# dialpattern-group 3 AP-GS1004(config-dialpattern-group#3)#	Create GSM Callback Group.
2	AP-GS1004(config-dialpattern-group#3)# pattern 0 2000	Enter GSM callback number.
3	AP-GS1004(config-dialpattern-group#0)# exit	Return to configuration mode.
4	AP-GS1004(config)# voice-port 1/0	Enter to Callback applicable port.
5	AP-GS1004(config-voice-port-1/0)# gsm-call-back-	Apply callback.



	white-list-group 3	
6	AP-GS1004(config-voice-port-1/0)# exit AP-GS1004(config)#	Exit.
7	AP-GS1004(config)# exit AP-GS1004#	Exit from input mode.



GSM Command [LCR(Least Cost Routing)]

```
• GSM - LCR(Least Cost Routing)
```

```
LCR Sample Configuration
!
hostname AP-GS3000
!
interface fastethernet 0/0
ip address 194.168.1.2 255.255.255.0
!
ip route 0.0.0.0 0.0.0.0 194.168.1.1
! Time interval Group configuration.
!
time-interval-group 0
 weekdays from 09:00 to 12:00
!
!
! LCR Tariff Group configuration.
!
Icr-tariff-group 0
 time-interval-group 0
  restore-limit daily
 accounting-period first 5 other 10
 free-quota voice 300
 free-quota sms 100
```

Step	Command	Explanation
1	AP-GS3000(config)# time-interval-group 0 AP-GS3000(config-tinterval-group#0)#	Create Time Interval Group.
2	AP-GS3000(config-tinterval-group#0)# weekdays from 09:00 to 12:00	Setup a date and time for LCR.
3	AP-GS3000(config-tinterval-group#0)# exit	Exit.



4	AP-GS3000(config)# lcr-tariff-group 0 AP-GS3000(config-lcr-group#0)#	Create LCR Tariff Group.
5	AP-GS3000(config-lcr-group#0)# time-interval- group 0	Apply previously setup Time Interval Group.
6	AP-GS3000(config-lcr-group#0)# restore daily	Setup Restore. (everyday, every month)
7	AP-GS3000(config-lcr-group#0)# accounting- period first 5 other 10	Create accounting regard of Voice Call.
8	AP-GS3000(config-lcr-group#0)# free-quota voice 300	Setup free phone call time
9	AP-GS3000(config-lcr-group#0)# free-quota sms 100	Setup free SMS number.
10	AP-GS3000(config-lcr-group#0)# exit AP-GS3000(config)#	Exit.



GSM Command [Rest]

• GSM - GSM Port Functions

Sample Configuration for GSM Band Selection
!
GSM 1/0
sim pin 1234
band-select gsm900
bts bcch 338
sms-language ascii
!
End

Step	Command	Explanation
1	AP-GS3000(config)# gsm 1/0 AP-GS3000(config-gsm1/0)#	Enter to GSM Port.
2	AP-GS3000(config-gsm1/0)# sim pin 1234	Setup PIN Code. (When necessary)
3	AP-GS3000(config-gsm1/0)# band-select gsm900	Setup GSM Bandwidth. (900 , 850, 1800, 1900)
4	AP-GS3000(config-gsm1/0)# bts bcch 338	Setup cell: Auto / Manual.
5	AP-GS3000(config-gsm1/0)# sms-language utf8	Setup SMS language.
6	AP-GS3000(config)# exit	Exit.

• GSM - SMS

Step	Command	Explanation
1	AP-GS3000# gsm 1 0 sms message send <phone number=""> <message></message></phone>	Send SMS from GSM Port.
2	AP-GS3000# gsm 1 0 message list all	Check SMS Message. (All / Unread / Read / Fail)
3	AP-GS3000# gsm 1 0 sms message delete all	Delete SMS Message. (All / Partially)

• GSM – USSD (Balance check of SIM Card)

Step	Command	Explanation
1	AP-GS3000# gsm 1 0 ussd send *123#	Check SIM Card Balance.

• GSM - Radius

Sample Configuration for Radius
!
radius-server host 192.168.100.254 1813 1812
radius-server key addpac
radius-server retransmit 3
radius-server timeout 5
radius-server authentication voip-inbound
radius-server authentication username 1234 password 1234
radius-server accounting voip start-stop
1 1
End

Step	Command	Explanation
1	AP-GS3000(config)# radius-server host <ip address=""> <accounting port=""> <authentication port=""></authentication></accounting></ip>	Setup Radius Server IP address, Auth port, Accounting port



2	AP-GS3000(config)# radius-server key addpac	Setup Radius Server Password		
3	AP-GS3000(config)# radius-server authentication voip- inbound / voip-outbound / voip-bidirectional	Setup VoIP call direction for Radius Authentication. Inbound / Outbound / Both		
4	AP-GS3000(config)# radius-server authentication username 1234 password 1234	Setup Authentication Username and password.		
5	AP-GS3000(config)# radius-server accounting voip start- stop / stop-only	Setup Accounting Type Start-Stop / Stop-only		
6	AP-GS3000(config)# radius-server retransmit 3	Number of retransmit time		
7	AP-GS3000(config)# radius-server timeout 5	Setup Timeout (sec)		



Command for GSM [Monitoring]

Step	Command	Explanation
1	AP-GS3000# show gsm sim-card	Check sim card status of GSM Gateway
2	AP-GS3000# show gsm module status	Check GSM module information and status
3	AP-GS3000# show gsm <slot> <port> cell-monitor</port></slot>	Display cell information to slot/port
4	AP-GS3000# show gsm <slot> <port> check-dev</port></slot>	Check device status to slot/port
5	AP-GS3000# show gsm <slot> <port> dev-info</port></slot>	Check device information to relevant slot/port (only be seen after debugging)
6	AP-GS3000# show gsm <slot> <port> imei</port></slot>	Check IMEI to relevant slot/port
7	AP-GS3000# show gsm <slot> <port> imsi</port></slot>	Check IMSI to relevant slot/port
8	AP-GS3000# show gsm <slot> <port> net-oper-id</port></slot>	Check GSM Network Operator ID to relevant slot/port
9	AP-GS3000# show gsm <slot> <port> phone-number</port></slot>	Check sim number to relevant slot/port
10	AP-GS3000# show gsm <slot> <port> reg-status</port></slot>	Check GSM registration status to relevant slot/port
11	AP-GS3000# show gsm <slot> <port> rssi</port></slot>	Check cell RSSI to relevant slot/port
12	AP-GS3000# show gsm <slot> <port> serv-cell-info</port></slot>	Display cell information to relevant slot/port
13	AP-GS3000# show gsm <slot> <port> state</port></slot>	Display operation status to relevant slot/port

• GSM - Monitoring command



•	GSM	- Debugging	command
---	-----	-------------	---------

Step	Command	Explanation
1	AP-GS3000# debug gsm <slot><port> all</port></slot>	Debug on regard of all GSM
2	AP-GS3000# debug gsm <slot><port> call</port></slot>	Debug on regarding GSM call
3	AP-GS3000# debug gsm <slot><port> cmd</port></slot>	Debug on regarding GSM command
4	AP-GS3000# debug gsm <slot><port> event</port></slot>	Debug on regarding GSM event
5	AP-GS3000# debug gsm <slot><port> error</port></slot>	Debug on regarding GSM error
6	AP-GS3000# debug gsm <slot><port> sms</port></slot>	Debug on regarding GSM SMS



Chapter 4. Appendix

Console Port Signal and Pin Out

In this appendix, it explains the cable pin specification of GS3000.

- Console port signal and pin out (RJ-45 to DB9)
- UTP cable (RJ-45 to RJ-45) pin out
- E1/T1 cable (RJ-45 to PBX) pin out

[Console Port Signal and Pin Out]

Use built-in RJ-45 to DB9 (Female DTE connector) to connect router console port and PC which terminal emulator software runs.

Console Port (DTE)	RJ-45	DB-9	Console Device (PC)
Signal	RJ-45 Pin	DB-9 Pin	Signal
RTS	1	8	CTS
DTR	2	6	DSR
TxD	3	2	RxD
GND	4	5	GND
GND	5	5	GND
RxD	6	3	TxD
DSR	7	4	DTR
CTS	8	7	RTS

<Chart 4-1> Console Port Pin Out

[UTP cable (RJ-45 to RJ-45) Pin Out]

Use RJ-45 to RJ-45 Ethernet cable to connect router and other devices (HUB). RJ-45 connector pin order is shown in Picture 4-1.



<Picture 4-1> 100Base-TX RJ-45 Connector

RJ-45	Signal	Direction	RJ-45 Pin
]	Tx +	\rightarrow	1
2	Tx -	\rightarrow	2
3	Rx +	\leftarrow	3
4	-	-	4
5	-	-	5
6	Rx -	←	6
7	-	-	7
8	-	-	8

<Chart 4-2> Series Ethernet Cable Signal and Pin Out

- 1. This specification is cable specification for series cable in between router and hub.
- 2. Use cross cable to directly connect router to PC or router to router.

[E1/T1 Cable (RJ-45 to PBX) Pin Out]

Addpac E1/T1 module can be connected with RJ45 cable. RJ45 pin arrangement is as below. Connect PBX RX with RJ45 1,2 and TX with 3,4.



<Picture 4-2> Digital E1/T1 RJ45 Pin Out





AddPac Technology

2F, Kyeong-An Bldg., 769-12 Yeoksam-Dong. Kangnam-gu, Seoul. 135-080 Phone (02) 568-3848 Fax (02) 568-3847 Email: sales@addpac.com Homepage: www.addpac.com

