IP Telephony Solution for Train Traffic Control Center (General Train)





AddPac Technology

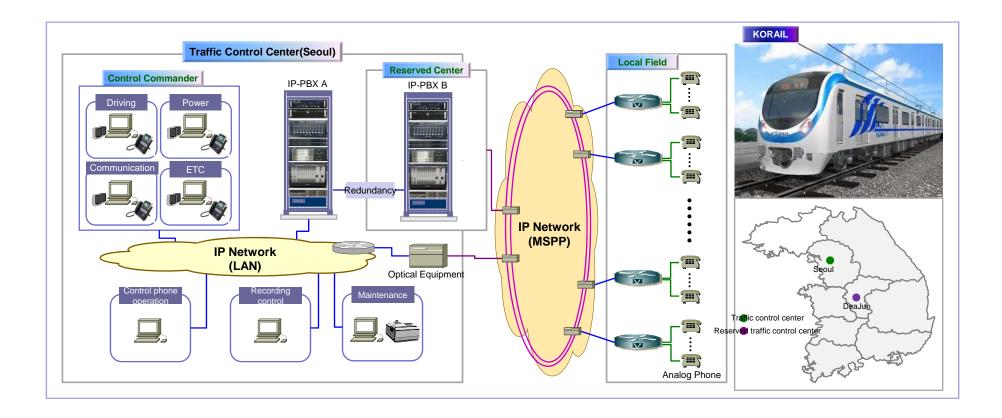
2012, Sales and Marketing

Contents

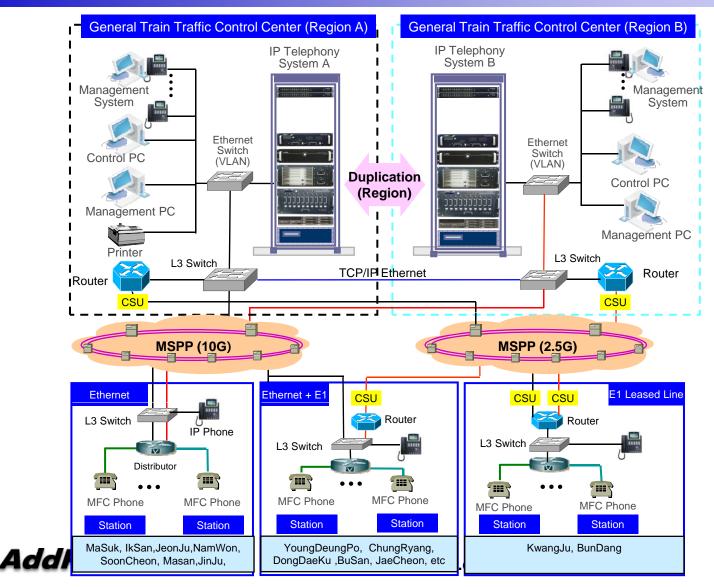
- IP Telephony Network Diagram for General Train Traffic Control Center (Korail)
- IP Telephony Solution Components (Example)
 - AP6800 Large Scale VoIP Gateway (Long Distance Support)
 - IPNext3000 Call Manager
 - AP-NR2000L IP Voice Recording Server
 - AP-MC1500 Voice MCU
 - AP2620 VoIP Gateway with Voice Recording Feature
 - AP-IP300 IP Phone
 - AP-PT50 Extend Speed-Dial Box
 - AP-SAC Smart Attendant Console
 - AP-SNMS Smart IP Telephony NMS



IP Telephony Network Diagram for Korail(cont..)



IP Telephony Network Diagram for Korail



IP Telephony Solution Components



AP6800 VoIP Gateway (Large Scale, Long Distance)



Large Capacity VoIP Gateways

Product	AP6500	AP6800
Available VoIP Modules	AP-N1-FXS32 AP-N1-FXO32	AP-N1-FXS32 AP-N1-FXO32
Analog Ports	Up to 128 (32-Port Module x 4)	Up to 256 (32-Port Module x 8)
Signaling	SIP, H.323	SIP, H.323
CPU Redundancy (Dual CPU)	Support (Option)	Support (Option)
E&M	N/A	N/A
Module Slot for VoIP Module	4 Slots	8 Slots
LAN Port	2	2
Console	1	1
Dual Power Supply (Option)	Support	Support



Main Features

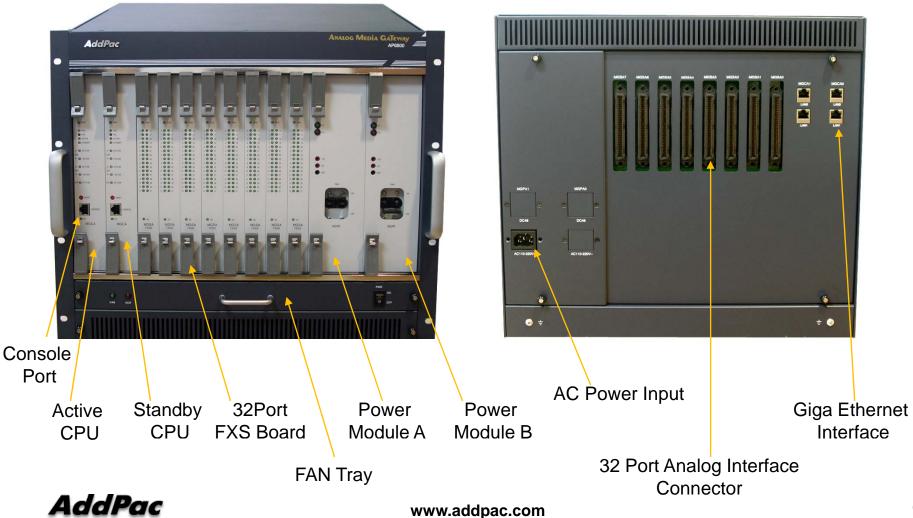
AP6800 Large Scale VoIP Gateway

- SIP/MGCP Concurrent Redundancy Stack Embedded
- High Performance RISC Processor
- Dual CPU Redundancy (Active, Standby) for System Stability
- Dual Power Redundancy (AC, DC) selectable at ordering
- Hot-Swap Function Support
- Up to 256 Analog Interface Port (32Port Board x 8 Ea) Support
- Two(2) 10/100/1000Mbps Giga-Bit Ethernet Interface
- G.711/G.726/G.723/G.729, T.38 Fax , VAD, etc
- Powerful Network Protocols (PPPoE, DHCP, Static Routing, etc)
- Firmware Upgradeable Architecture
- VPMS (VoIP Plug&Play Management System) for Large Scale Deployment
- Advanced Voice QoS Mechanism



AP6800 Large Scale VoIP Gateway

RISC CPU High-end DSP



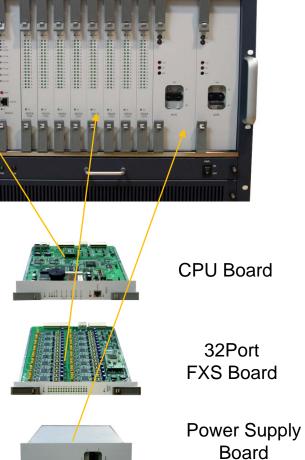
AP6800 Large Scale VoIP Gateway

Hardware Specifications

AP6800 VoIP Gateway	Basic Specifications	
Dual Redundancy CPU	High-End RISC Microprocessor,	
	Dual Redundancy CPU (Active, Standby)	
Voice Interface	Up to 256-Port FXS Voice Interface	
	Up to 256-Port FXO Voice Interface	
	32-Port Analog FXS Board, FXO Board	
Ethernet Interface	2-Ports 10/100/1000Mbps Giga Ethernet	
	Interface(RJ-45)	
Flash Memory/CPU	8Mbyte High-speed Flash Memory	
Base Memory/CPU	64/128Mbyte High-speed SDRAM	
Power Requirement	Dual Power Supply / VAC 110~220V,	
	50/60Hz, 880Watt	
Operating Temperature	0°C ~ 45°C (32 °F ~ 122°F)	
Storage Temperature	-40°C ~ 85°C (-40°C ~ 185°F)	
Relative Humidity	5% ~ 95% (Non-condensing)	
Dimensions	400mm x 482mm x 340mm (H x W x D)	



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IPNext3000 Call Manager





Main Features

IPNext3000 Call Manager

- IP based Advanced PBX Solution
- IPv4/IPv6 based Dual Network Protocol Support
- Internal/External RTP Proxy Function Support
 - External RTP Proxy Server for Private Address : AP-RS2000
- Internal/External Presence Function Support with Smart Messenger
 - External Presence Server : AP-PS2000
- Powerful Management and User Friendly Features
- Fault Tolerant and Scalability Architecture
- High-performance Video, Audio, and Voice Service
- Firmware Upgradeable Architecture
- Dual System Redundancy Architecture
 - Two(2) 3.5 Inch Hard Disk (RAID 1) / System
 - Two(2) Gigabit Ethernet Interface / System
- Dual Redundancy Power Module



IPNext3000 Call Manager

- 64bit High-End Microprocessor Computing Power
- Main Chassis
 - Dual Redundancy CPU Boards for System Fault Tolerant
 - Two(2) 10/100Mbps Gigabit Ethernet
 - One(1) RS-232C Console (RJ45)
 - Two(2) 3.5 Inch Hard Disk Interface Slot (RAID 1)
 - Dual Redundancy Power Supply Module
 - Hot-Swap Features





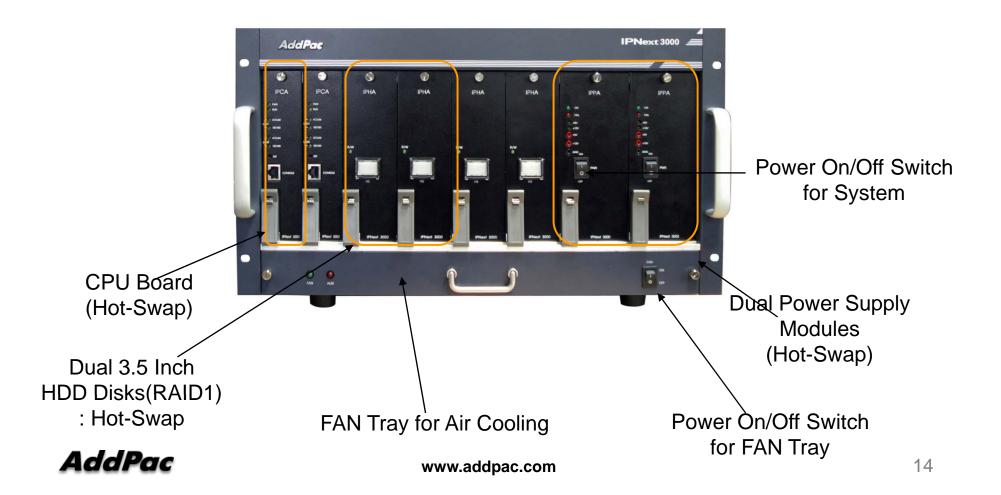
RISC

CPU

Hardware Specification IPNext3000 Call Manager

^{RISC}

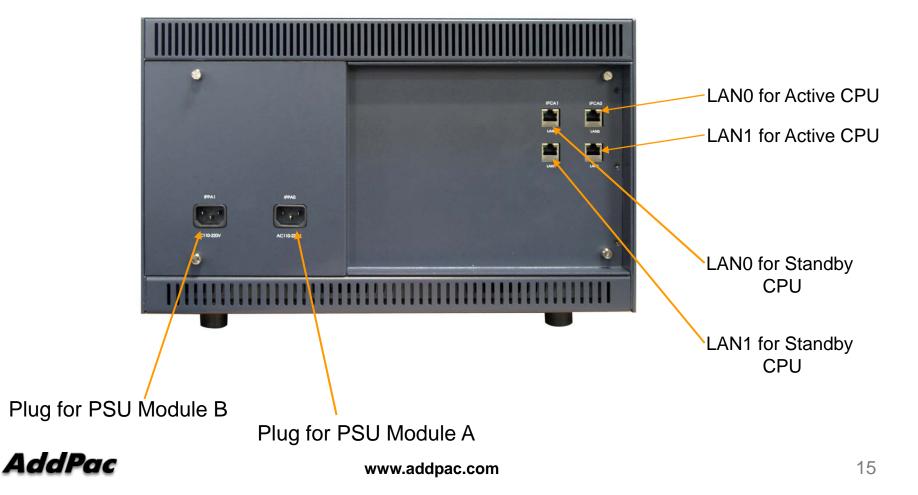
IPNext3000 Front Side



Hardware Specification IPNext3000 Call Manager

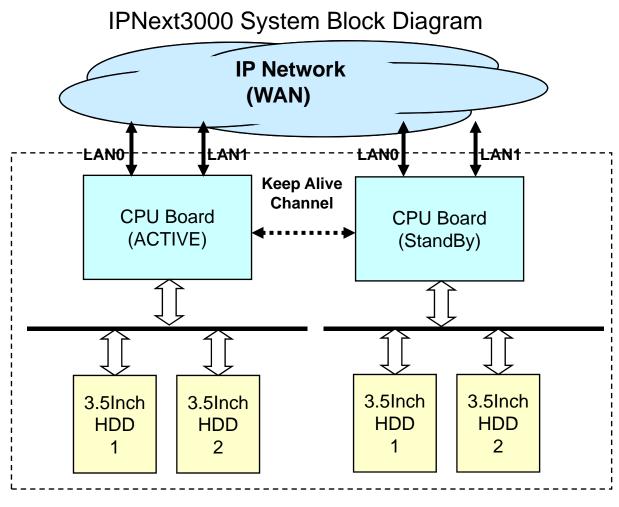


IPNext3000 Back Side



System Redundancy Features

IPNext3000 Call Manager

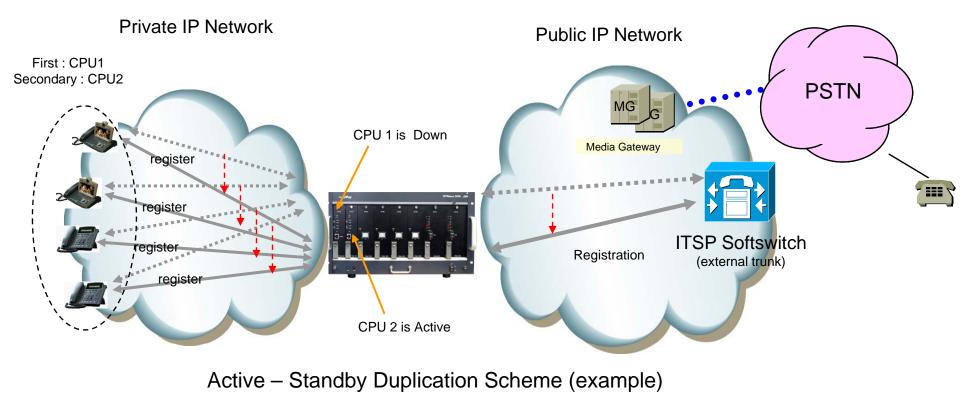


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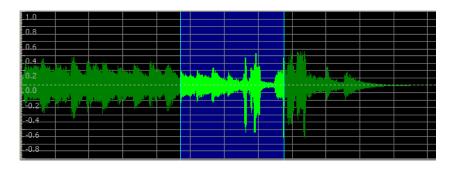
System Redundancy Features

IPNext3000 Call Manager

- Active-Active Duplication Scheme
- Active Standby Duplication Scheme
- VRRP based Duplication Scheme



AP-NR2000L IP Voice Recording Server





Main Features

AP-NR2000L IP Voice Recoding Server

- IP based Network Voice Recording Server
- Powerful Management and User Friendly Features
- Fault Tolerant and Scalability Architecture
- High-performance Voice Recoding Service
- External Voice Transcoder(ex:AP-VTC1000) Interworking Support
- Firmware Upgradeable Architecture
- Two(2) Gigabit Ethernet Interface
- Ten(10) 3.5Inch IDE type Hard Module Slot
- RAID 1 Hard Disk Backup
- Smart Digital Voice Recording Management Software

AP-NR2000L IP Voice Recoding Server

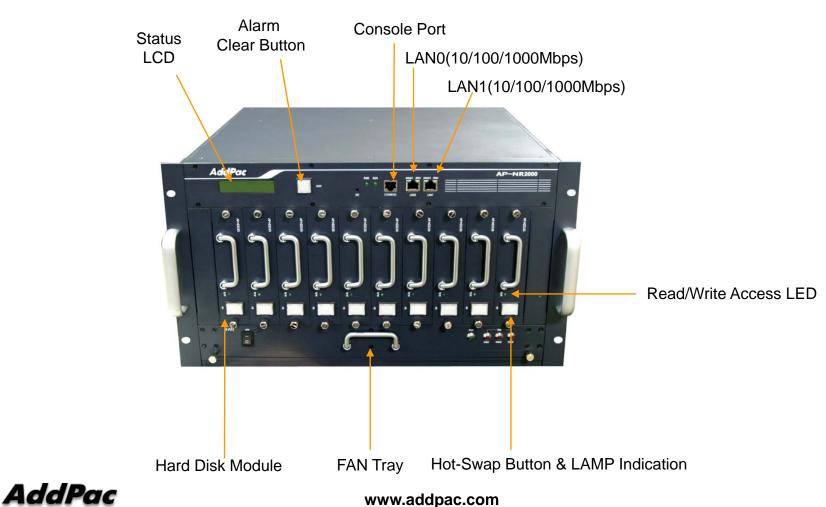


- **RISC Microprocessor Computing Power**
- Main Chassis
 - Network Interface
 - Two(2) 10/100/1000Mbps Gigabit Ethernet
 - One(1) RS-232C Console (RJ45)
 - Ten(10) Hard Disk Interface Slot for RAID1 Configuration



AP-NR2000L IP Voice Recoding Server





Front Side

AP-NR2000L IP Voice Recoding Server



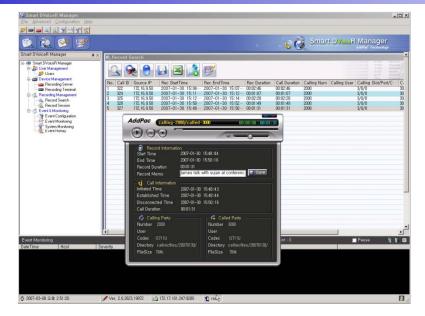
Power Supply Module Power ON/OFF Power Inlet Switch

Back Side



Smart DV_{oice}R Management Program

AP-NR2000L IP Voice Recoding Server

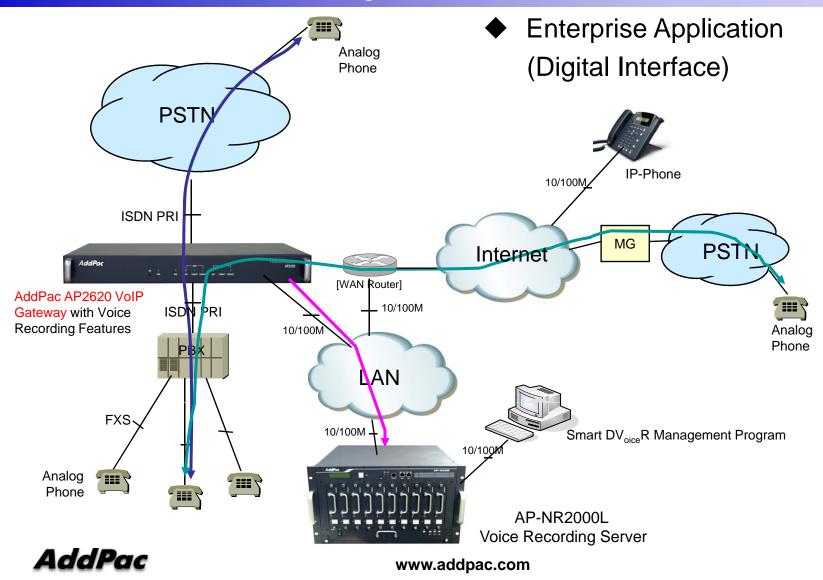


- Call History Management (search/modify/delete/save)
- Media Play Management (Play/Stop/Seek/Pause)
- Live Call List Management, Live Call Monitoring
- Local Backup (File Manager Support, PC HDD, DVD) and Local Play
- User Management (registration/modify/delete/search)
- Server Status (CPU/Memory/HDD) & Event Monitoring
- Waveform Analyzing Function
- Recording Source Management (VoIP Gateway, IP Phone, etc)
- Live Recording Board



Network Service Diagram

AP-NR2000L IP Voice Recoding Server



AP-MC1500 Audio MCU



Main Features

AP-MC1500 Voice MCU

- IP based Hardware Audio Mixing Solution
- Two(2) DSP Module Slots for Audio Mixing
- 64Ch, 128ch Audio Mixing DSP Modules
- Compact PCI Style Hot-Swap Support
- Support Various Conference System Model
- Scalability and Flexibility
- High-performance Voice Mixing Service
- SIP and H.323 VoIP Signaling Support
- Various Voice Codec Support (G.711,G.729, G.723, etc)
- Two(2) 10/100Mbps Fast Ethernet Interface
- Firmware Upgradeable Architecture
- Audio Solution with Outstanding Network Service Capability



AP-MC1500 Audio MCU

- High Performance RISC Microprocessor Computing Power
- Built-in High-end Programmable DSP Hardware Architecture
- Main Chassis
 - Network Interface
 - Two(2) 10/100Mbps Fast Ethernet
 - One(1) RS-232C Console (RJ45)





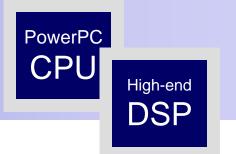
PowerPC

CPU

High-end

DSP

AP-MC1500 Audio MCU







AP-MC1500 Audio MCU

- MCU Module (HIM-AMCU128)
 - High quality Audio Mixing
 - 128Channel Audio Mixing
 - Compact PCI Style Hot-Swap Function
 - High-End Programmable DSPs
 - Parallel DSP Processing for High Quality Audio Mixing
 - Concurrent Different Audio Codec Support
- MCU Module (HIM-AMCU64)
 - High quality Audio Mixing
 - 64 Channel Audio Mixing
 - Compact PCI Style Hot-Swap Function
 - High-End Programmable DSPs
 - Parallel DSP Processing for High Quality Audio Mixing
 - Concurrent Different Audio Codec Support
 - Cascade Support





AP2620 VoIP Gateway with Voice Recording Feature



VoIP Gateways with Voice Recording Extension

Product	AP2620	AP2640	AP2650	AP-MG3000
	Addres - 1 - 1 FIL FIL - Marting		· · · · · · · · · · · · · · · · · · ·	
Available Modules	AP-FXS4 AP-FXO4 AP-FXS2O2 AP-E&M4 AP-E1	AP-FXS8 AP-FXO8 AP-FXS4O4 AP-E&M8 AP-E1	AP-FXS8 AP-FXO8 AP-FXS4O4 AP-E&M8 AP-E1	APv2-1E1 APv2-2E1 APv2-4E1
Analog Ports	Up to 8	Up to 32	Up to 32	N/A
Signaling	SIP, H.323	SIP, H.323	SIP, H.323	SIP, H.323
Digital E1/T1	Up to 2E1/T1	Up to 2E1/T1	Up to 2E1/T1	Up to 4E1/T1
E&M	Support	Support	Support	N/A
Module Slot	Two(2)	Four(4)	Four(4)	Two(2)
LAN Port	2	2	2	2
Console	1	1	1	1
Power	Single PSU	Single PSU	Dual PSU	Single PSU

Main Features

AP2620 VoIP Gateway

- Voice Recording Interface Support
- FXS, FXO, E&M, Digital E1/T1 VoIP Interface
- Two(2) Module Slots for Analog/Digital VoIP Interface
- Up to 2E1/T1 Digital Interface
- Up to 8 Port Analog Interface
- IPv4/IPv6 Address Support
- SIP, H.323, MGCP VoIP Signaling Support
- Firmware Upgradeable Architecture
- Rack Mountable Chassis









Main Features

- 4.3 Inch Color LCD Display
- External I/O Interface
 - Audio In/Out
 - Two(2) Fast Ethernet Interface
 - PSTN FXO Interface (optional)
- PoE (Power over Ethernet) Support
- 25 Speed Dial button with Presence Indication Lamp
- Audio Broadcasting Controller & Terminal
- Providing Powerful Push-to-Talk Service
- Powerful Color GUI
- IPv4/IPv6 Address Support
- SIP, H.323 Signaling Support
- High-end Error Resilient Against Various Packet Error



- RISC+DSP Microprocessor Computing Power (Dual Processor Architecture)
- High Quality 4.3 Inch Color LCD Panel
- 25 Speed Dial Key & User Presence Indication LED
- Optional PSTN Backup Interface
 - FXO Interface
- High quality Audio and Voice Interface
 - Stereo Audio Input Connector
 - Stereo Audio Output Connector
- Network Interface
 - Two(2) 10/100Mbps Fast Ethernet
- USB Host Mode Interface
 - USB Memory(Flash, HDD), etc
- Power Supply
 - Power over Ethernet
 - External Power Adaptor (5V, 3A)













IP Extend Key Pack Comparison Table

Model	AP-PT100	AP-PT50	AP-PT20		
Service Features					
Кеу Туре	7 inch LCD Touch Screen	Push Button with User Presence Indication	Push Button with User Presence Indication		
		LAMP	LAMP		
Key Number	Default : 9(row) x 4(column) = 36	60 Key	40 Key		
User Presence Indication	Support	LED on, LED off, LED Blink	LED on, LED off, LED Blink		
Multiple Cascading	Support	Support	Support		
Speaker	Support	Support	Support		
LAN Port	2	2	2		
PoE(Option)	Support	Support	Support		
Application	IP Phone or Video Phone Extend Key Pack	IP Phone or Video Phone Extend Key Pack	IP Phone or Video Phone Extend Key Pack		



Main Features

AP-PT50 IP Extend Key Pack

- Speed Dialing Terminal with User Presence Indication LAMP
- 60 (3 Column x 20 Row) Speed-Dial Key Buttons
- Inter-working with IP Terminal, Presence Server, Call Manager
- Multiple Terminal Cascading Feature
- User Presence Indication LED (on, off, blink)
- User Presence Group Feature
- Firmware Upgradeable Architecture
- PoE (Power over Ethernet) Function



Hardware Specification

AP-PT50 IP Extend Key Pack

- RISC+DSP Microprocessor Computing Power
- 60 Speed Dial Key with User Presence LAMP
- Optional PoE (Power over Ethernet) Interface
- Network Interface
 - Two(2) 10/100Mbps Fast Ethernet
- Power Supply
 - Power over Ethernet (Option)
 - External Power Adaptor (5V, 2A)
 - Power Switch
- Connector for Cascading





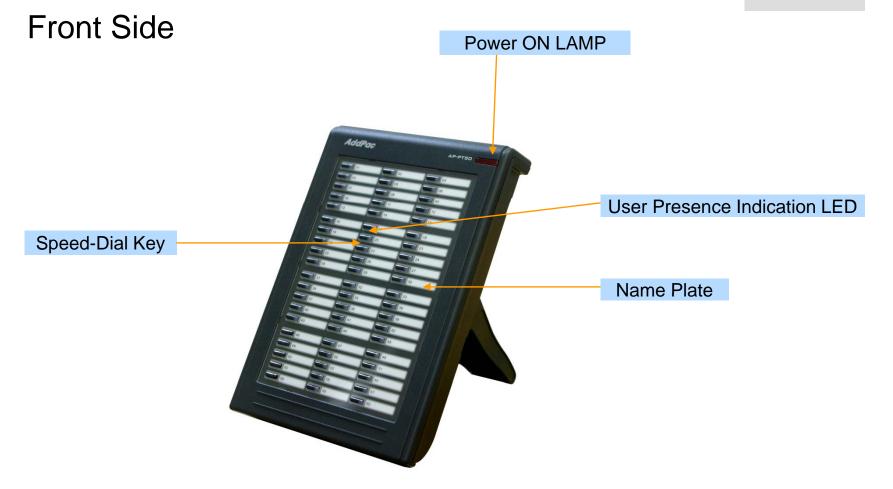
RISC

CPU

Hardware Specification

AP-PT50 IP Extend Key Pack

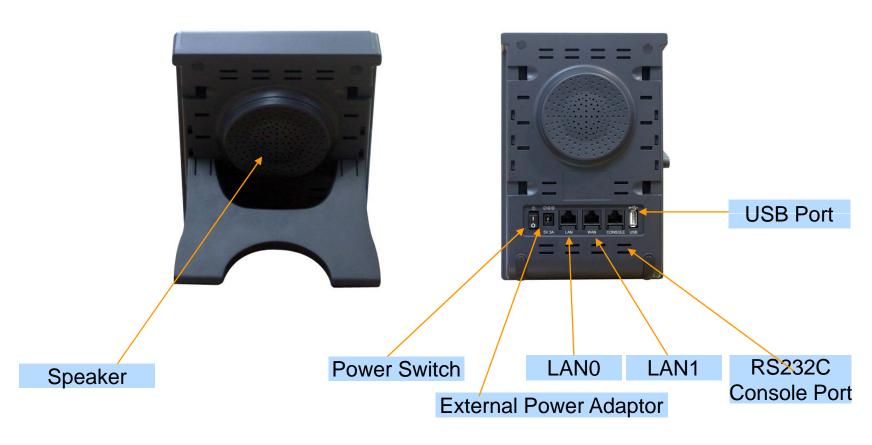
RISC CPU



Hardware Specification

AP-PT50 IP Extend Key Pack

Back Side





RISC

CPU

AP-SAC Smart Attendant Console (Touch Screen Version)



Contents

- Overview
- Touch Screen based Attendant Console System Diagram
- Touch Screen based Attendant Console Feature
- Touch Screen based Attendant Console UI
 - Login Feature
 - Main Screen Feature

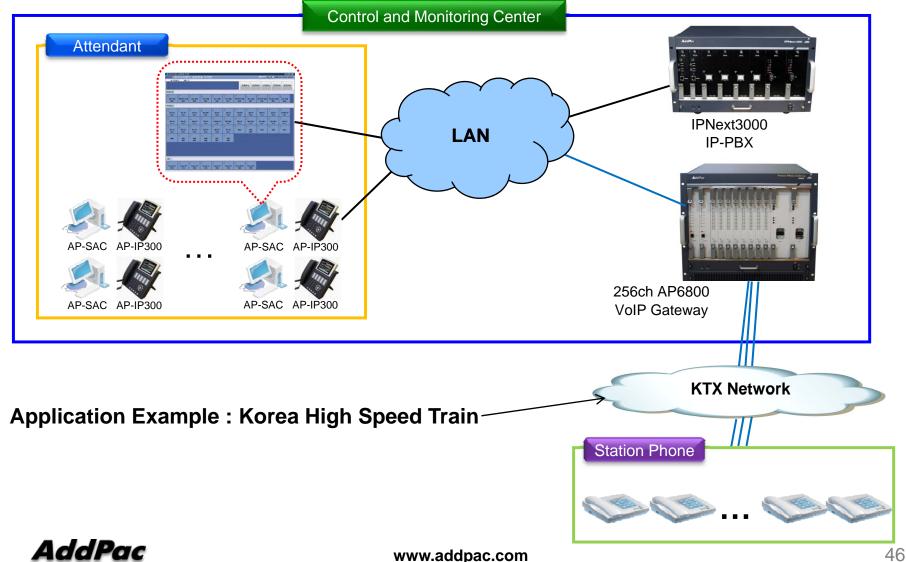
Overview

Support Efficiently Call Routing and Smart Presence Service

- Touch Screen based Attendant Console Function
 - Support Call Routing Service
 - Support User Presence Information
 - Support Phone Number Presence Information
 - Support Attendant Phone Call Control
 - Support Call Status Display
 - Support Group Call(Conference)
 - Support Various Monitoring Group



Touch Screen based Attendant Console System Diagram



Touch Screen based Attendant Console Feature

- User Login Service
 - User Login and Logout Feature
- Monitoring and Group Management Service
 - Support Various Group
 - Attendant and Direct Phone Group
 - Office Phone Group
 - Conference Call Group
 - Support Group Management Feature
 - Add Group, Rename Group, Delete Group
 - Add User, Delete User
 - Add Conference Room, Delete Conference Room



Touch Screen based Attendant Console Feature

- Monitoring and Group Management Service
 - Presence Feature
 - Phone Presence Status
 - Online, Working, Away, Absence
 - Be In Conference, Call Busy
 - User Presence Status
 - PC Online, Phone Online, PC-Phone Online
 - Working, Away, Absence
 - Be In Conference, Call Busy
 - Display Call Status
 - Display Call Status Icon for Talking Time
 - Call Control Feature
 - Interoperate with Monitoring Phone Information
 - Manual Call Dial
 - Drag and Drop Call Transfer and Manual Call Transfer



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Touch Screen based Attendant Console Feature

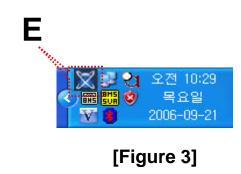
- Attendant Phone Call Control Service
 - Display Phone Call Status
 - Display Call Status Icon for Call Stat
 - Display Talking Call Status Icon for Talking Time
 - Support Multi-Call
 - Phone Call Control Feature
 - Dial to Monitoring Phone and Conference
 - Incoming Call Answer
 - Active Call Hold and UnHold
 - Call Transfer
 - Call Disconnect
 - Interoperate with Monitoring Phone Information
 - Drag and Drop Call Transfer and Manual Call Transfer



Touch Screen based Attendant Console User Login

	Attendani	t Console	
I D	cmc	Login	
Passwo	ord ****	Config	
🗖 Auto log	gin 🔽 Save passwo	ord Vesrion:1.0.5.0	
	[Fig	jure 1] •	
Ĺ	<i>,</i>		
B Sëtting::::	127.0.0.1	×	

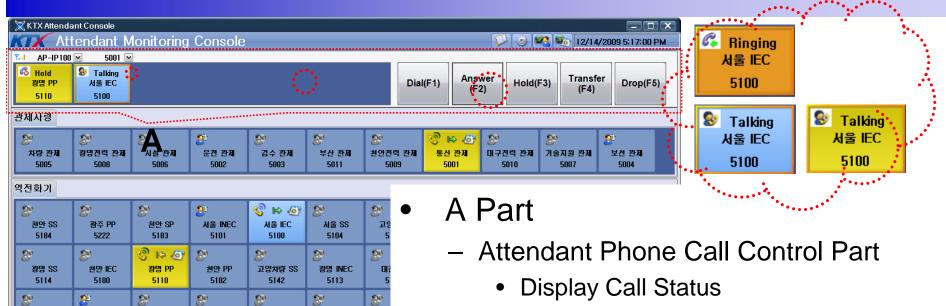




- Attendant Console Login
 - [Figure 1] shows the login picture of KTX Attendant Console program. "E" of [Figure3] is Icon of KTX Attendant Console Program.
 - When program is started initially, user should set the network configuration parameter such as IP address, Port number of Presence Server.[Figure2]
 - After inputting ID and Password, press the login button



Touch Screen based Attendant Console Main Screen – Call Control



- Support Multi-Call Display
- Support Call Control Feature
 - Dial, Answer
 - Hold/UnHold
 - Transfer
 - Drop
- Support Short Key Call Control



광주 SS

5224

6006

20

그룹

20

Group SS

7010

대전 SS

5194

6003

6003

200

200

Group BD

7170

광주 IEC

5220

6004

6004

Ŷ

90

Group SP

7020

대전 PP

5192

6000

6000

20

20

Group CA

7180

대전 IEC

5190

6001

6001

20

29

Group PP

7030

광주 SP

5223

6002

6002

20

20

Group DJ

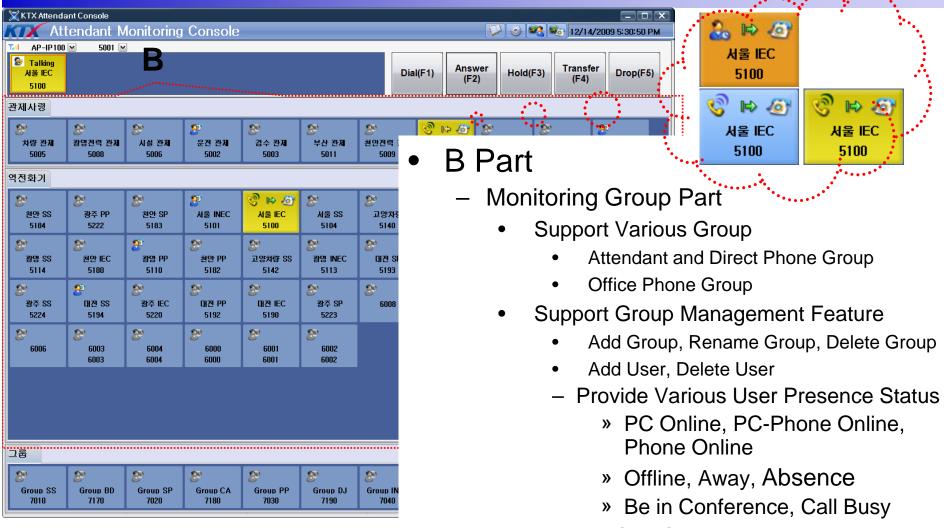
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Touch Screen based Attendant Console Main Screen – Monitoring

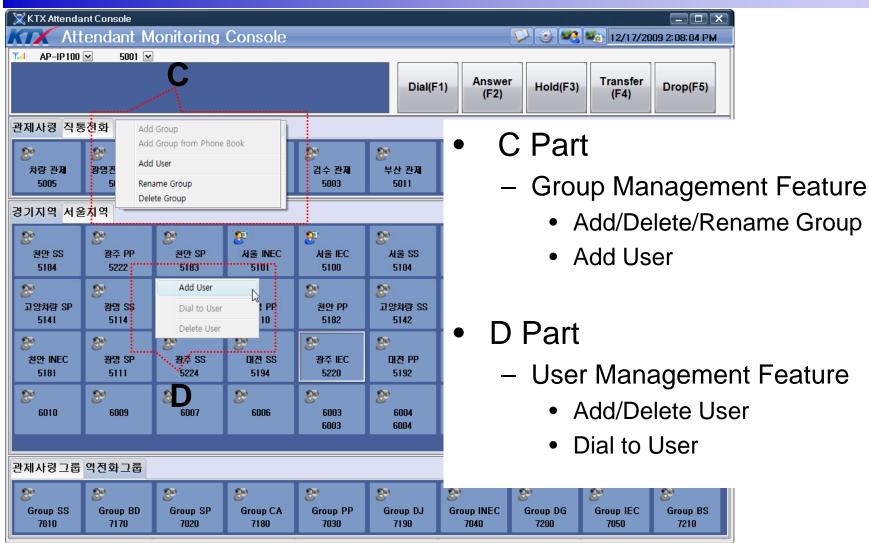


Display Call Status



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Touch Screen based Attendant Console Main Screen – Monitoring Group





Touch Screen based Attendant Console Main Screen – Group Call

KTX Attendant Console Attendant Monitoring Console AP-IP100 S001 S001						
관제사령 직통	전화					E Part
29	8	8	<u></u>	8	29	 Group Call(Conference) Part
차량 관제 5005	광명전력 관제 5008	시설 관제 5006	운전 관제 5002	김수 관제 5003	부산 50	 Support Various Group
경기지역 서울지역					 Conference Call Group 	
운 천만 SS 5184	광주 PP 5222	운 천안 SP 5183	운 서울 INEC 5101	운 서울 IEC 5100	원 전물 51	F Part
고양차량 SP 5141	운왕 광명 SS 5114	운행 천만 IEC 5180	종 광명 PP 5110	천만 PP 5182	設 고양大 51	 Support Group Management Feature
운 천안 INEC	왕명 SP	양 광주 SS 5224	운 한 대전 SS 5194	응 광주 IEC 5000	201 대전 51	 Add Conference Room, Delete Conference Room
5181	5111 8009	5224 6007	6006	5220 6003 6003	60 60	 Add Conference Group, Rename Conference Group, Delete Conference Group
관제사령그룹	역전화그룹	Add Group				
Sroup SS 7010	Sroup BD 7170	Add User Rename Grou Delete Group	1.00	Group PP 7030	Sroup DJ 7190	Provide Group INEC 7040Provide

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Smart Network Management System



Main Features

- Web-based Management
- Network Resource Management
- Device Fault Management
- Device Fault History Management
- Device Status Information
- Notification Management
- Fault Statistics
- Model & Service Management



Thank you!

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