

SmartLink M-ATA Micro-Analog Telephone Adapter

Getting Started Guide



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Firmware Revision Information

This version of the Getting Started Guide is based on **firmware revision (1026)**.

Summary Table of Contents

1	General information	17
2	SmartLink installation.....	19
3	Home.....	22
4	Network—LAN	25
5	Telephony	32
6	System	48
7	Documentation	68
8	Logout.....	70
9	Contacting Patton for assistance	72
A	Compliance information	75
B	Specifications	77
C	Dial plans	80
D	Calling Features	83
E	Voice prompt configuration	85

Table of Contents

Summary Table of Contents	3
Table of Contents	4
List of Figures	12
List of Tables	13
About this guide	14
Audience.....	14
Structure.....	14
Precautions	15
Safety when working with electricity	15
General observations	16
Typographical conventions used in this document.....	16
General conventions	16
1 General information	17
SmartLink M-ATA overview	18
2 SmartLink installation	19
Installing the M-ATA.....	20
Resetting the M-ATA to factory default	21
3 Home	22
System information	23
System Uptime	23
LAN IP Address	23
MAC address	23
Application Version	23
Config Date	23
Security	24
Application Code Version	24
Downloader Code Version	24
System Status.....	24
SIP Messages Sent	24
SIP Messages Received	24
SIP Bytes Sent	24
SIP Bytes Received	24
RTP Packets Sent	24
RTP Packets Received	24
RTP Bytes Sent	24
RTP Bytes Received	24
4 Network—LAN	25
Status.....	26
Interface Status	26

Enabled	26
Service	26
Interface Status	26
Network Settings	26
Dynamic IP Assignment	26
IP address	27
MAC address	27
Subnet Mask	27
Default Gateway	27
Domain name	27
DNS address	27
DynDNS address	27
VLAN	27
Priority Tag	27
Settings.....	27
Internet Configuration	28
Obtain LAN configuration dynamically	28
Specify fixed LAN configuration	28
WAN PPPoE Configuration	29
Enable PPPoE	29
Authentication	29
Settings	29
Idle Timeout	29
Echo Timeout	29
Echo Count	29
Service Name	29
AC Name	29
Dynamic DNS	29
Configuring Dynamic DNS	30
MAC Spoofing Configuration	30
WAN MAC Address (Spoofed)	30
VLAN Configuration	30
VLAN Tag (IEEE 802.1q)	31
Priority Tag (IEEE 802.1q)	31
Saving your work	31
ToS/DiffServ.....	31
Saving your work	31
5 Telephony	32
VoIP Status	35
VoIP Server Registration Status	35
Current Server	35
Domain	35
Base RTP Port	35

Phone Line Status	35
Registration Status	35
User Name	36
Caller ID Setting	36
Subscribed for Voicemail	36
Messages waiting	36
SIP	36
SIP Configuration	37
SIP Server Settings	37
Gateway Settings	37
Dial Plan	37
SIP Extensions	38
Support PRACK method	38
Encode SIP URI with user parameter	38
Send INVITE with Timer header	38
Call Hold using C=0.0.0.0	38
Send NOTIFY	38
RTP Telephone Event Configuration	38
VoIP VLAN Configuration	38
SIP Parameters	39
Hook Flash MIME Type	39
SIP Timer Values (milliseconds)	39
SIP T1	39
SIP T2	39
SIP T4	39
RTP Parameters	39
NAT Traversal	40
Outbound Proxy IP	40
Outbound Proxy Port	40
Stun Server IP	40
Stun Server Port	40
UPnP	40
NONE	40
Saving your work	40
Audio/CODEC Configuration.....	41
CODECS	41
Packetization	41
Jitter Buffer	41
FAX without T.38 (Use G.711 fax)	41
Saving your work	42
Phone 1	42
User Information	42
Phone Number	43
User Name	43

Port	43
CallerID Name	43
Password	43
Supplementary Service Activation	43
Call Forward All	43
Selective Call Forward	43
Three-Way Conferencing	43
Incoming Call Block	43
Distinctive Ring	44
Call Transfer	44
Dialing by IP Address	44
Speed Dial	44
Message Waiting Indicator	44
Call Forward on Busy	44
Conditional Call Forward	44
Call Waiting	44
Anonymous Call Reject	44
Caller ID	44
Call Return	44
Do Not Disturb	44
Self Caller ID Block	44
Outgoing Call Block	44
Dial Out Type	45
Dial Out Type	45
Hot Line Number	45
Warm Line Number	45
Call Forward Settings	45
Cfwd All Dest	45
Cfwd Busy Dest	45
Selective Call Forward Settings	46
Incoming caller #1–8	46
Forward destination #1–8	46
Incoming Call Block	46
Block Caller ID	46
Outgoing Call Block	46
Digit Pattern	46
HTTP Digest Setting	47
Saving your work	47
Speed Dial	47
Line 1 Speed Dial Settings	47
Speed Dial Serv	47
Speed Dial 1–8 Phone Number/IP Dialing	47
Saving your work	47

6 System	48
Set Security Password	51
Web Page Protect	51
New Root Password	51
New User Password	51
Confirm new password	51
Saving your work	51
Configuration	52
Syslog Service	52
Enable Syslog	52
Syslog Server	52
Debug Service	53
Enable Debug	53
Debug Server	53
Debug Connection Port	53
NTP Service	53
NTP Server	54
Time Zone	54
Adjust clock for daylight savings	54
Saving your work	54
Service Access	54
Enable HTTP	54
Enable Ping Reply	54
Saving your work	54
Localization	55
Call Progress Tones	55
Dial Tone	56
Prompt Tone	56
Confirm Tone	56
Holding Tone	56
Busy Tone	56
Ring Back Tone	56
Off Hook Warning	56
Distinctive Ring Settings	57
Supplementary Service Keys	58
Call forward All	58
Call forward on Busy	58
Call forward selective	58
Call Waiting	58
Incoming Call Block	58
Block Anonymous Calls	58
Distinctive Ring	58
Warm Line	58
Do Not Disturb	59

IP Dialing	59
Speed Dialing	59
Income Caller ID	59
Self Caller ID Block	59
Calling Prefix Keys	59
Call Return	59
Warm Line	59
Speed Dial	60
Call Forward All	60
IP Dialing	60
Call Hold	60
Call Waiting (call alternative)	60
Call Conference	60
Call Conference Drop	60
Call Transfer	60
Voicemail access	61
Call Forward Busy Destination	61
Hotline	61
Control Timer Values	61
Hook Flash Timer	61
SIP Session Timer value	61
Conditional Call Forwarding Timer	62
Warm Line Delay	62
Interdigit Timer	62
Offhook Idle Time	62
Offhook Warning tone time	62
FXS Port	62
FXS port Input Gain	62
FXS port Output Gain	62
Caller ID Method	62
“Call Progress Tones” Synchronization	62
Ring Setting	63
Ring Waveform	63
Ring Frequency	63
Ring Voltage	63
FXS Port Polarity Configuration	63
Idle Polarity	63
Caller Conn Polarity	63
Saving your work	63
SNMP Configuration	64
SNMP Trap Configuration	64
IP address	64
Trap Community	64
SNMP Community Configuration	64

Read Community	64
Write Community	64
SNMP System Configuration	64
System Description	64
System Object Id	64
Saving your work	64
Auto Upgrade.....	65
Routine Upgrade every xxx day(s)	65
Enable Auto Upgrade	65
Auto Upgrade Protocol	65
Upgrade Server	65
Auto Upgrade URL	65
Saving your work	65
Manual Upgrade.....	66
Reload	67
7 Documentation	68
Introduction	69
8 Logout.....	70
Introduction.....	71
9 Contacting Patton for assistance	72
Introduction.....	73
Contact information.....	73
Patton support headquarters in the USA	73
Alternate Patton support for Europe, Middle East, and Africa (EMEA)	73
Warranty Service and Returned Merchandise Authorizations (RMAs).....	73
Warranty coverage	74
Out-of-warranty service	74
Returns for credit	74
Return for credit policy	74
RMA numbers	74
Shipping instructions	74
A Compliance information	75
Compliance.....	76
EMC Compliance:	76
Safety Compliance	76
Radio and TV Interference (FCC Part 15)	76
CE Notice (Declarations of Conformity).....	76
B Specifications	77
Voice Connectivity.....	78
Connectivity.....	78
Voice Processing (signalling dependent)	78
Fax and Modem Support.....	78

Voice Services/Features	79
IP Services	79
Management	79
Operating Environment	79
System	79
C Dial plans	80
Introduction	81
Sample Dial Plans.....	81
Simple Dial Plan	81
Non-dialed Line Dial Plan	81
Complex Dial Plan	81
D Calling Features	83
Introduction.....	84
E Voice prompt configuration	85
Introduction	86
Accessing the voice prompt	86
Existing voice prompt configuration	86

List of Figures

1	SmartLink M-ATA	18
2	SmartLink M-ATA installation diagram	20
3	SmartLink VoIP download and configuration Home page	23
4	Internet Status window	26
5	Internet Configuration section of the Settings window	28
6	WAN PPPoE Configuration section of the Settings window	29
7	Dynamic DNS Service window	30
8	MAC Spoofing Configuration section of the Settings window	30
9	VLAN Configuration section of the Settings window	31
10	TOS/DiffServ window	31
11	VoIP Status window	35
12	SIP Configuration section of the SIP window	37
13	Gateway Settings section of the SIP window	37
14	SIP Extensions section of the SIP window	38
15	RTP Telephone Event Configuration section of the SIP window	38
16	VoIP VLAN Configuration section of the SIP window	39
17	SIP Parameters section of the SIP window	39
18	NAT Traversal section of the SIP window	40
19	Audio/CODEC Configuration window	41
20	User Information section of Phone 1 window	42
21	Supplementary Service Settings section of Phone 1 window	43
22	Dial Out Type section of Phone 1 window	45
23	Call Forward Settings section of Phone 1 window	45
24	Selective Call Forward Settings section of Phone 1 window	46
25	Incoming Call Block section of Phone 1 window	46
26	Incoming Call Block section of Phone 1 window	46
27	HTTP Digest Setting section of Phone 1 window	47
28	Speed Dial window	47
29	Set Security Password window	51
30	Configuration window	52
31	Service Access window	54
32	Call Progress Tones section of Localization window	56
33	Distinctive Ring Setting section of Localization window	57
34	Supplementary Service Keys section of Localization Window	58
35	Calling Prefix Keys section of Localization Window	59
36	Control Timer Values section of Localization window	61
37	FXS Port Polarity Configuration section of Localization window	62
38	Ring Setting section of Localization window	63
39	FXS Port Polarity section of Localization window	63
40	SNMP Configuration window	64
41	AutoUpgrade window	65
42	Manual Upgrade window	66
43	Reload window	67
44	Documentation link	69
45	Logout window	71
46	Password verification page	71

List of Tables

1	General conventions	16
2	Bellcore standard ring cadence patterns	57
3	Calling features	84

About this guide

This guide describes using the SmartLink M-ATA.

Audience

This guide is intended for the following users:

- VoIP telephony service providers
- Enterprise telecom and IT technicians



Consumers of VoIP telephony service providers and employees of enterprises that received the Patton smartLink in conjunction with VoIP services are encouraged to contact their provider before making any configuration changes. Improper configuration may lead to a loss of services.

Structure

This guide contains the following chapters and appendices:

- [Chapter 1](#) on page 17 provides information about M-ATA features and capabilities
- [Chapter 2](#) on page 19 provides hardware installation procedures
- [Chapter 3](#) on page 22 describes the Home section settings
- [Chapter 4](#) on page 25 describes the LAN section settings
- [Chapter 5](#) on page 32 describes the Telephony section settings
- [Chapter 6](#) on page 48 describes the System section settings
- [Chapter 7](#) on page 68 describes how to download and display the SmartLink M-ATA user guide
- [Chapter 8](#) on page 70 describes how to log out of the M-ATA management system
- [Chapter 9](#) on page 72 contains information on contacting Patton technical support for assistance
- [Appendix A](#) on page 75 contains compliance information for the M-ATA
- [Appendix B](#) on page 77 contains specifications for the M-ATA
- [Appendix C](#) on page 80 describes dialing plans and contains sample plans
- [Appendix D](#) on page 83 describes the calling features that can be accessed from phones attached to the SmartLink M-ATA
- [Appendix E](#) on page 85 explains how to configure the voice prompt function in the SmartLink M-ATA

For best results, read the contents of this guide *before* you install the M-ATA.

Precautions

Notes, cautions, and warnings, which have the following meanings, are used throughout this guide to help you become aware of potential problems. **Warnings** are intended to prevent safety hazards that could result in personal injury. **Cautions** are intended to prevent situations that could result in property damage or impaired functioning.

Note A note presents additional information or interesting sidelights.



IMPORTANT

The alert symbol and IMPORTANT heading calls attention to important information.



CAUTION

The alert symbol and CAUTION heading indicate a potential hazard. Strictly follow the instructions to avoid property damage.



CAUTION

The shock hazard symbol and CAUTION heading indicate a potential electric shock hazard. Strictly follow the instructions to avoid property damage caused by electric shock.



WARNING

The alert symbol and WARNING heading indicate a potential safety hazard. Strictly follow the warning instructions to avoid personal injury.



WARNING

The shock hazard symbol and WARNING heading indicate a potential electric shock hazard. Strictly follow the warning instructions to avoid injury caused by electric shock.

Safety when working with electricity



WARNING

The SmartLink contains no user serviceable parts. The equipment shall be returned to Patton Electronics for repairs, or repaired by qualified service personnel.



WARNING

The external power adapter shall be a listed Limited Power Source. Ensure that the power cable used with this device meets all applicable standards for the country in which it is to be installed, and that it is connected to a wall outlet which has earth ground. The mains outlet that is utilized to power the device shall be within 10 feet (3 meters) of the device, shall be easily accessible, and protected by a circuit breaker.



Do not work on the system or connect or disconnect cables during periods of lightning activity.



In accordance with the requirements of council directive 2002/96/EC on Waste of Electrical and Electronic Equipment (WEEE), ensure that at end-of-life you separate this product from other waste and scrap and deliver to the WEEE collection system in your country for recycling.

General observations

- Clean the case with a soft slightly moist anti-static cloth
- Place the unit on a flat surface and ensure free air circulation
- Avoid exposing the unit to direct sunlight and other heat sources
- Protect the unit from moisture, vapors, and corrosive liquids


Typographical conventions used in this document

This section describes the typographical conventions and terms used in this guide.

General conventions

The procedures described in this manual use the following text conventions:

Table 1. General conventions

Convention	Meaning
Garamond blue type	Indicates a cross-reference hyperlink that points to a figure, graphic, table, or section heading. Clicking on the hyperlink jumps you to the reference. When you have finished reviewing the reference, click on the Go to Previous View button  in the Adobe® Acrobat® Reader toolbar to return to your starting point.
Garamond bold type	Indicates the names of command buttons that execute an action.
< >	Angle brackets indicate function and keyboard keys, such as <SHIFT>, <CTRL>, <C>, and so on.

Chapter 1 **General information**

Chapter contents

SmartLink M-ATA overview	18
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SmartLink M-ATA overview

The SmartLink Micro-Analog Telephone Adapter (see [figure 1](#)) provides transparent connectivity for analog phones and faxes to the world of Internet voice. Connecting to any analog phone, fax or PBX, the SmartLink product is an effective and flexible solution for accessing Internet-based telephone services and corporate intranet systems across established LAN and Internet connections like xDSL and cable modems.



Figure 1. SmartLink M-ATA

The SmartLink M-ATA provides one RJ-45 Ethernet port and one FXS (RJ-11) analog phone port. Side panel LEDs quickly show at-a-glance the status of the system, WAN and phone ports.

A full suite of IP features (DHCP and NAT/PAT) are available on the M-ATA to provide easy interconnection to VoIP services and IP transport networks. VLAN tagging and prioritization enables voice traffic to be handled before data traffic by other devices on the network.

The web interface offers two levels of configuration access for the network operator and end user. The friendly web interface and product labeling (Phone, LAN, and WAN) to help ensure a trouble-free installation for the end user. Configuration and firmware can be downloaded from a TFTP server or HTTP server.

Chapter 2 **SmartLink installation**

Chapter contents

Installing the M-ATA	20
Resetting the M-ATA to factory default	21

Installing the M-ATA

- 1 The unit should be installed in a dry environment with at least 2 inches (5 cm) of clearance at the sides, front, and rear of the unit to allow air circulation for cooling.
- 2 Plug in the telephone (see figure 2).
- 3 Plug in the PC or LAN, or a LAN hub/switch.
- 4 Plug the power adapter into the power jack on the SmartLink M-ATA (see figure 2). Connect the other end of the power cord to an appropriate AC power outlet.
- 5 Wait 30 seconds after powering the SmartLink M-ATA on, then verify that the green *Power* LED is lit (see figure 2).
- 6 By default, the M-ATA will automatically request IP network settings from the LAN using DHCP. To determine the IP address of the SmartLink, lift the handset off the attached analog phone and dial * * * *.

Dial **1 0 0 #**, listen to and record the IP address of the SmartLink. (To manually set the IP address, see appendix E, “Voice prompt configuration” on page 85 for details).

- 7 Use a web browser to connect to the SmartLink M-ATA. The URL will be `http://<ip address>`. For example, if the M-ATA IP address was `10.10.10.2`, the URL would be `http://10.10.10.2`.

Note The default password for the M-ATA is “root”.



Follow the directions of your voice service provider to set up voice services.

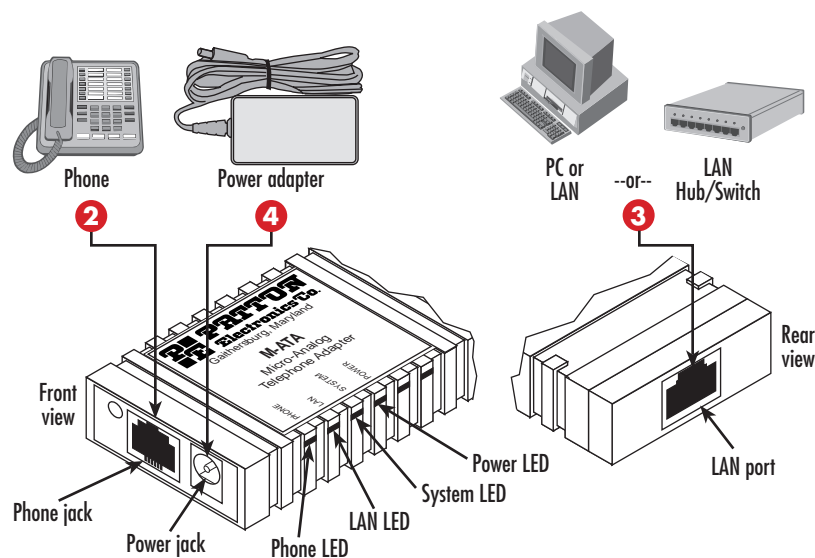


Figure 2. SmartLink M-ATA installation diagram

Resetting the M-ATA to factory default

- 1 The M-ATA should be powered off.
- 2 Press the recessed button on the rear of the M-ATA, then apply power to the unit.
- 3 Continue depressing the button while the LEDs are flashing, and only release the button after the LEDs have stopped.
- 4 You may now log in to the M-ATA using the password “root”.

Chapter 3 **Home**

Chapter contents

System information	22
System Uptime	22
LAN IP Address	22
MAC address	22
Application Version	22
Config Date	22
Security	23
Application Code Version	23
Downloader Code Version	23
System Status.....	23
SIP Messages Sent	23
SIP Messages Received	23
SIP Bytes Sent	23
SIP Bytes Received	23
RTP Packets Sent	23
RTP Packets Received	23
RTP Bytes Sent	23
RTP Bytes Received	23

System information

The screenshot shows the web interface for the SmartLink Micro - ATA. The page title is "SmartLink Micro - ATA" and the URL is "www.patton.com". A navigation menu on the left includes options like Home, Network - LAN, Status, Settings, ToS, Telephony, VoIP Status, SIP, CODECS, Phone 1, Speed Dial, System, Security, Configuration, Localization, SNMP, Auto Upgrade, Manual Upgrade, Reload, Documentation, User's Guide, and Logout. The main content area displays a welcome message and two sections: "System Information" and "System Status".

System Information

System Uptime:	0 days, 3h 25m 19s
NTP time:	03:27PM 12/27/2005 (GMT-8) DST
LAN IP Address:	10.10.22.10 (Dynamic)
MAC Address:	00:a0:ba:01:a0:6d
Application Version:	
Config Date:	
Security:	Password installed
Application Code Version:	SIP version 4.01.001 OE EN MA (1220)
Downloader Code Version:	3.1 EN (OEMA 011103)

System Status

SIP Messages Sent:	0	SIP Bytes Sent:	0
SIP Messages Recv:	0	SIP Bytes Recv:	0
RTP Packets Sent:	0	RTP Bytes Sent:	0
RTP Packets Recv:	0	RTP Bytes Recv:	0

Figure 3. SmartLink VoIP download and configuration Home page

System Uptime

Shows how long the SmartLink M-ATA has been operating since the last time it was reloaded (either by powering the unit off and then on again, or by selecting *System > Reload*).

LAN IP Address

The IP address of the SmartLink M-ATA. If *(Static)* is shown next to the address, it means the IP address was assigned to the SmartLink M-ATA under the LAN Settings. *(Dynamic)* indicates the IP address was learned via DHCP.

MAC address

The media access control (MAC) address of the Ethernet interface in the SmartLink.

Application Version

The SmartLink firmware version number.

Config Date

The date of the configuration file that was downloaded from an auto-upgrade server.

Security

Indicates that the SmartLink web interface utility has been secured with a password. To configure a password, see section “[Set Security Password](#)” on page 51.

Application Code Version

Shows the application code version being used.

Downloader Code Version

Shows the downloader code version being used.

System Status

Shows VoIP statistics for the period of time since the SmartLink M-ATA was last reloaded (either by powering the unit off and then on again, or by selecting *System > Reload*).

SIP Messages Sent

Total number of VoIP SIP messages sent (including retransmissions).

SIP Messages Received

Total number of VoIP SIP messages received (including retransmissions).

SIP Bytes Sent

Total number of bytes of VoIP SIP messages sent (including retransmissions).

SIP Bytes Received

Total number of bytes of VoIP SIP messages received (including retransmissions).

RTP Packets Sent

Total number of VoIP RTP packets sent (including redundant packets).

RTP Packets Received

Total number of VoIP RTP packets received (including redundant packets).

RTP Bytes Sent

Total number of VoIP RTP bytes sent.

RTP Bytes Received

Total number of VoIP RTP bytes received.

Chapter 4 **Network – LAN**

Chapter contents

Status.....	25
Interface Status	25
Enabled	25
Service	25
Interface Status	25
Network Settings	25
Dynamic IP Assignment	25
IP address	26
MAC address	26
Subnet Mask	26
Default Gateway	26
Domain name	26
DNS address	26
DynDNS address	26
VLAN	26
Priority Tag	26
Settings.....	26
Internet Configuration	27
Obtain LAN configuration dynamically	27
Specify fixed LAN configuration	27
WAN PPPoE Configuration	28
Enable PPPoE	28
Authentication	28
Settings	28
Idle Timeout	28
Echo Timeout	28
Echo Count.....	28
Service Name	28
AC Name.....	28
Dynamic DNS	28
Configuring Dynamic DNS	29
MAC Spoofing Configuration	29
WAN MAC Address (Spoofed)	29
VLAN Configuration	29
VLAN Tag (IEEE 802.1q)	30
Priority Tag (IEEE 802.1q)	30
Saving your work	30
ToS/DiffServ	30
Saving your work	30

Status

Shows the status of key LAN network settings as configured under *LAN settings*

Interface Status

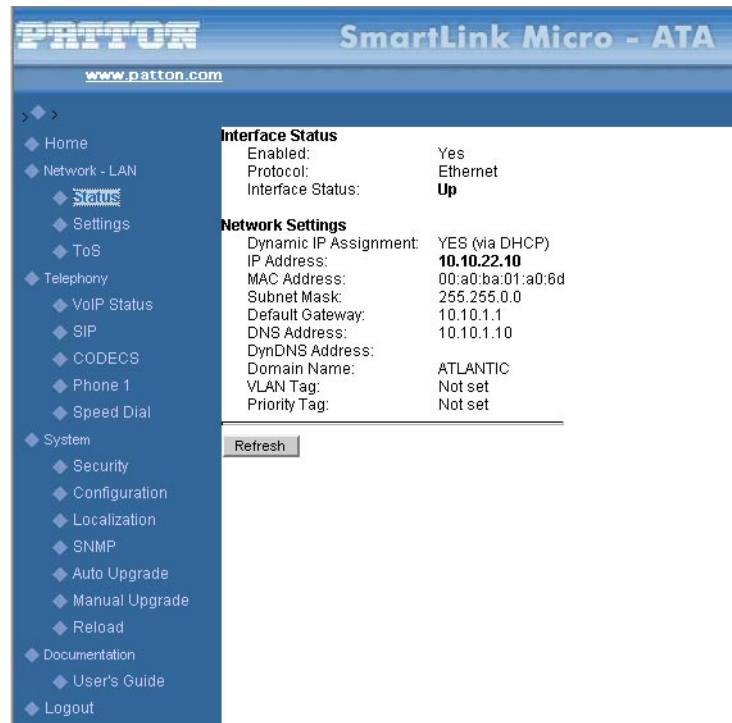


Figure 4. Internet Status window

Enabled

Yes indicates the LAN interface is enabled and ready to be used.

Service

Either *Routed* or *Bridged*, displays whether the SmartLink's LAN interface connection is operating in a routed or bridged mode.

Interface Status

Either *Up* or *Down*.

Network Settings

These are the details of your LAN network settings.

Dynamic IP Assignment

Displays *Yes (via DHCP)* if you are using a dynamic IP address or *No* if a dynamic IP address is not being used.

IP address

The IP address of the SmartLink on the WAN interface.

MAC address

The MAC address of the WAN Ethernet interface in the SmartLink.

Subnet Mask

The subnet mask is 32-bit number that filters a destination IP address to determine to which subnet it belongs. For example, a subnet mask of *255.255.0.0* for a network ID of *192.5.0.0* tells the switch to accept traffic destined for IP addresses that begin with *192.5*—all other packets are ignored.

Default Gateway

The IP address of the gateway. The gateway IP address can be retrieved automatically in DHCP mode or be set up manually with a fixed IP address.

Domain name

The network domain name of the SmartLink.

DNS address

Refers to the address of your domain name server that was defined under LAN settings or that was learned dynamically through DHCP.

DynDNS address

The IP address of the dynamic DNS server that will be notified when the SmartLink M-ATA's dynamic IP address changes.

VLAN

VLAN tag value encoded in the LAN Ethernet header in all outgoing packets

Priority Tag

Priority tag value encoded in the LAN Ethernet header in outgoing packets.

Settings

The *Settings* window contains the following sections:

- Internet Configuration (see [figure 5](#) on page 28)
- WAN PPPoE Configuration (see [figure 6](#) on page 29)
- MAC Spoofing Configuration (see [figure 8](#) on page 30)
- Internet VLAN Configuration (see [figure 8](#) on page 30)

Note After configuring the sections, click the **Save Internet Settings** button (see [figure 4](#) on page 26) to save the new configuration.

Internet Configuration

Figure 5. Internet Configuration section of the Settings window

Obtain LAN configuration dynamically

Select this option if appropriate. If you choose *Obtain LAN configuration dynamically*, the information is detected automatically through DHCP.

Specify fixed LAN configuration

Select this option if you will not be using DHCP. If you choose *Specify fixed LAN configuration*, you will have to enter the following information:

- IP address.
- IP of the netmask.
- IP of the gateway.
- IP of the DNS Server, if applicable.
- Host name (the name will identify the computer on the Internet, such as *SL4020.patton.com*).
- Domain name (the name that will identify one or more IP addresses). For example, the *patton.com* domain is used by Patton Electronics Company. That domain can include multiple hostnames (such as *SL4020.patton.com*, *ftp.patton.com*, and so on) that point to individual computers on the Patton network. In short, for the hostname URL *http://www.patton.com*, the domain name is *patton.com*.

WAN PPPoE Configuration

The screenshot shows a configuration window titled "WAN PPPoE Configuration". It is organized into two main sections: "Authentication" and "Settings".

- Authentication:** Contains two text input fields: "Username:" and "Password:".
- Settings:** Contains five input fields: "Idle Timeout:(minutes)", "Echo Timeout:(seconds)", "Echo Count:", "Service Name:", and "AC Name:".

Figure 6. WAN PPPoE Configuration section of the Settings window

Enable PPPoE

Select *Yes* to enable PPPoE or *No* to disable PPPoE.

Authentication

Enter the username and password provided by your ISP.

Settings

Idle Timeout. Idle timeout before PPP connection is closed due to inactivity

Echo Timeout. The duration between sending PPP echo requests to server.

Echo Count. The number of unanswered PPP echo requests before the PPP connection is closed.

Service Name. PPPoE Service name

AC Name. PPPoE access concentrator (AC) name

Dynamic DNS

The SL4020 supports Dynamic DNS for use in environments where the IP address of LAN IP interface is not assigned statically (i.e. permanently) but instead is assigned dynamically using protocols like DHCP.

With a statically assigned IP addresses, DNS is used to establish a fixed relationship between an IP address a DNS name (example: 209.22.110.3 = joe@patton.com). The DNS name allows the IP device, like a phone attached to an SL4020, to be found by the DNS name (joe@patton.com) anywhere on the Internet.

Prior to Dynamic DNS, dynamically assigned IP address changes could not be automatically reflected in DNS. As a result, VoIP devices like the SL4020 could not always be located on the Internet by DNS name. This is especially true for cable modem and ADSL services that use DHCP to assign addresses to customers.

With Dynamic DNS, the SmartLink M-ATA will inform the dynamic DNS server of its IP address when it receives a new or dynamically assigned IP address from the network. This allows the SL4020 to always be found using a fixed DNS name when the IP address changes. With Dynamic DNS cable modem and ADSL users can be found on the Internet using a DNS name.

The DNS server used for registration is operated by Dynamic Network Services, Inc., Dynamic DNS (DynDNS). You can find detailed information about the company and the services it offers on the webpage www.dyndns.org. The company offers different levels of service. The basic services are offered free of charge, while the more advanced services are fee-based.

Figure 7. Dynamic DNS Service window

Configuring Dynamic DNS

1. Select *System > Configuration*.
2. **Choose Sever:** To enable DynamicDNS, select the name of the DYNDNS service from the drop-down menu (see [figure 7](#)).
3. **Host Name:** The host name is the name of the SL4020 as registered on the DYNDNS service.
4. **Username:** The user name is the user name as registered on the Dynamic DNS service.
5. **Password:** The password is the password as registered on the Dynamic DNS service.

MAC Spoofing Configuration

Figure 8. MAC Spoofing Configuration section of the Settings window

WAN MAC Address (Spoofed)

Only available when the unit is using the M-ATA mode. The spoofed MAC address to be used by the device's LAN interfaces, the Ethernet address of the outgoing packets from the LAN interface would be replaced with this address. If blank, the LAN interfaces will use the hardware value of MAC

VLAN Configuration

The SmartLink M-ATA can mark outgoing Ethernet frames on the LAN interface with VLAN and priority tags. Other devices on the LAN can use the tags to control how frames from the SmartLink M-ATA are pro-

cessed. All data leaving the SmartLink M-ATA will be marked with the specific value unless overridden on the telephony VLAN configuration settings.

The screenshot shows a web interface titled 'VLAN Configuration'. It contains two input fields: 'VLAN Tag:' and 'Priority Tag:'. Below these fields is a button labeled 'Save and Reload'.

Figure 9. VLAN Configuration section of the Settings window

VLAN Tag (IEEE 802.1q)

IEEE 802.1q Ethernet VLAN tag for all outgoing packets on LAN Ethernet interface. The value should be between 0 and 4094.

Priority Tag (IEEE 802.1q)

IEEE 802.1q Ethernet Priority tag for all outgoing packets on LAN Ethernet interface. The value should be between 0 and 7.

Saving your work

When you are finished configuring the VLAN settings, click the **Save and Reload** button (see [figure 9](#)) to save all changes.

ToS/DiffServ

This sub-page is used to configure the Type-of-Service/Diffserv byte values which are to be used in the IP header of all transmitted SIP signaling packets and RTP packets. The ToS/DiffServ byte values are entered as two-digit hexadecimal values. If no special ToS/DiffServ value is to be used for a particular traffic type, enter **00** or leave the setting empty.

The screenshot shows the 'ToS/DiffServ' configuration window. It has a left-hand navigation menu with options like Home, Network - LAN, Status, Settings, ToS, Telephony, and VoIP Status. The main content area has two input fields: 'Call Signalling Packets:' and 'RTP Packets:'. Both fields contain the value '54' and are followed by '(2 Hex digit byte value)'. Below these fields is a button labeled 'Save ToS/DiffServ Settings'.

Figure 10. ToS/DiffServ window

Saving your work

When you are finished configuring ToS/DiffServ settings, click the **Save ToS/DiffServ Settings** button to save the changes.

Chapter 5 **Telephony**

Chapter contents

VoIP Status	35
VoIP Server Registration Status	35
Current Server	35
Domain	35
Base RTP Port	35
Phone Line Status	35
Registration Status	35
User Name	36
Caller ID Setting	36
Subscribed for Voicemail	36
Messages waiting	36
SIP	36
SIP Configuration	37
SIP Server Settings	37
Gateway Settings	37
Dial Plan	37
SIP Extensions	38
Support PRACK method	38
Encode SIP URI with user parameter	38
Send INVITE with Timer header	38
Call Hold using C=0.0.0.0	38
Send NOTIFY	38
RTP Telephone Event Configuration	38
VoIP VLAN Configuration	38
SIP Parameters	39
Hook Flash MIME Type	39
SIP Timer Values (milliseconds)	39
SIP T1.....	39
SIP T2.....	39
SIP T4.....	39
RTP Parameters	39
NAT Traversal	40
Outbound Proxy IP	40
Outbound Proxy Port	40
Stun Server IP	40
Stun Server Port	40
UPnP	40
NONE	40
Saving your work	40

Audio/CODEC Configuration.....	41
CODECS	41
Packetization	41
Jitter Buffer	41
FAX without T.38 (Use G.711 fax)	41
Saving your work	42
Phone 1	42
User Information	42
Phone Number	43
User Name	43
Port	43
CallerID Name	43
Password	43
Supplementary Service Activation	43
Call Forward All	43
Selective Call Forward	43
Three-Way Conferencing	43
Incoming Call Block	43
Distinctive Ring	44
Call Transfer	44
Dialing by IP Address	44
Speed Dial	44
Message Waiting Indicator	44
Call Forward on Busy	44
Conditional Call Forward	44
Call Waiting	44
Anonymous Call Reject	44
Caller ID	44
Call Return	44
Do Not Disturb	44
Self Caller ID Block	44
Outgoing Call Block	44
Dial Out Type	45
Dial Out Type	45
Hot Line Number	45
Warm Line Number	45
Call Forward Settings	45
Cfwd All Dest	45
Cfwd Busy Dest	45
Selective Call Forward Settings	46
Incoming caller #1-8	46
Forward destination #1-8	46
Incoming Call Block	46
Block Caller ID	46

Outgoing Call Block	46
Digit Pattern	46
HTTP Digest Setting	47
Saving your work	47
Speed Dial	47
Line 1 Speed Dial Settings	47
Speed Dial Serv	47
Speed Dial 1–8 Phone Number/IP Dialing	47
Saving your work	47

VoIP Status

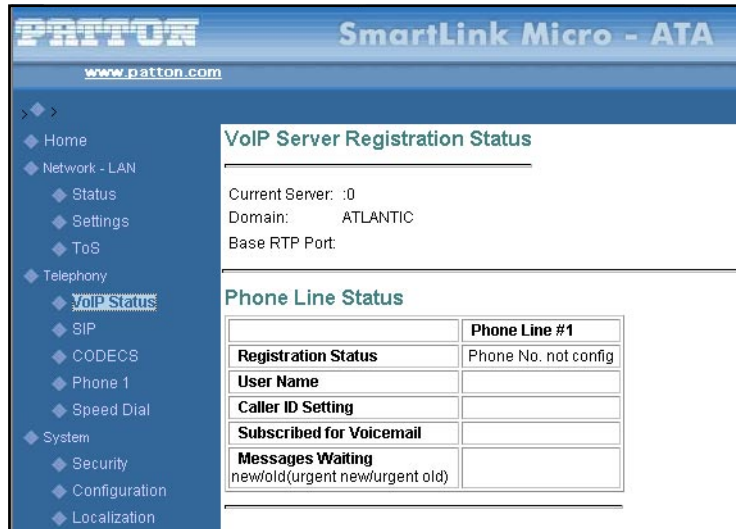


Figure 11. VoIP Status window

VoIP Server Registration Status

Current Server

Shows the current VoIP server that has been pre-defined or has been discovered using DNS service records (DNS-SRV).

Domain

The VoIP domain name is the domain name that is hosting the VoIP server.

Base RTP Port

Displays the base RTP port number for the RTP-RTCP port pair.

Phone Line Status

The *Phone Line Status* table shows the current operational status of the phone. It displays the VoIP registration status, the configured VoIP username, if the username is subscribed to voicemail and has messages waiting, and the caller ID setting. The messages waiting will indicate the number of new messages, old messages, new urgent messages and old urgent messages.

Registration Status

Shows the currently operational status of the phone. The following registration status values may be displayed:

- **Phone No. not config**—The phone number has not yet been configured for the phone line. Go to *Phone 1* to enter the phone configuration.
- **Online**—The phone line is online, registered with the SIP server and ready to send and receive phone calls using the SIP server.
- **Registering**—The SmartLink M-ATA is in the process of registering with the VoIP server.

- **No Registration**—The SmartLink M-ATA is not attempting to register with the SIP server because “Send Registration Request” is not checked on SIP sever setting screen.
- **Error: Not Authorized**—A SIP final status message status of 401 or 407 has been received from the SIP server.
- **Error: Forbidden**—SIP final status message status of 401 or 407 has been received from the SIP server.
- **Error 408: Request timeout**— A SIP final status message status of 408 has been received from the SIP server.
- **Response: xxx**—A SIP final status message status of xxx has been received from the SIP server.

User Name

The authentication username that was specified under section “Phone 1” on page 42. This user name will be used to register with the VoIP server.

Caller ID Setting

The Caller ID name setting that was specified under section “Phone 1” on page 42. This is the display name that others will see as your caller id when you make a call.

Subscribed for Voicemail

Shows if the SIP Subscribe message was successful for this phone line. If “Yes”, then the subscription process was successful. The number of voicemail messages in the queue will not be provided if the subscription for voice mail fails.

Messages waiting

After successful subscription to the SIP server, the SIP server/Internet Telephony service provider should send the number of voicemail messages that are queued up for the user using the SIP Notify message. The messages waiting shows the number of normal priority new messages, normal priority old (listened to) messages, number of urgent priority new voicemail messages and the number of old (listened to) urgent priority voicemail messages as reported to the SmartLink M-ATA by the VoIP SIP server.

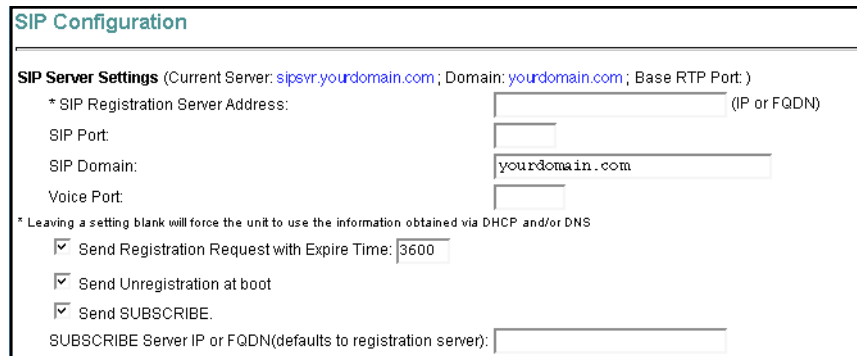
SIP

The *SIP* window contains the following sections:

- SIP Configuration (see [figure 12](#) on page 37)
- SIP Extensions (see [figure 14](#) on page 38)
- RTP Telephone Event Configuration (see [figure 15](#) on page 38)
- VoIP VLAN Configuration (see [figure 16](#) on page 39)
- SIP Parameters (see [figure 17](#) on page 39)
- NAT Traversal (see [figure 18](#) on page 40)

Note After configuring the sections, click the **Save SIP Settings** button (see [figure 18](#) on page 40) to save the new configuration.

SIP Configuration



SIP Configuration

SIP Server Settings (Current Server: sipsvr.yourdomain.com; Domain: yourdomain.com; Base RTP Port:)

* SIP Registration Server Address: [] (IP or FQDN)

SIP Port: []

SIP Domain: yourdomain.com

Voice Port: []

* Leaving a setting blank will force the unit to use the information obtained via DHCP and/or DNS

Send Registration Request with Expire Time: 3600

Send Unregistration at boot

Send SUBSCRIBE.

SUBSCRIBE Server IP or FQDN(defaults to registration server): []

Figure 12. SIP Configuration section of the SIP window

SIP Server Settings

The SmartLink M-ATA will automatically attempt to locate the VoIP server by using the domain name specified in the LAN interface or the server will be discovered via DHCP on the WAN interface. When found, the discovered server will be listed as the *Current Server*.

Enter the following information:

- Server address—The IP address or domain name hosting the VoIP SIP server
- Port—The UDP port of the VoIP SIP server. The default is 5060.
- Domain name—The VoIP domain name (realm) is used for validation of each phone's username
- Send Registration Request with Expire Time—If selected, determines the amount of time (in seconds) that the SIP registration will be valid for.
- Unregistration—If checked, the SmartLink will send a SIP unregister at system reload before sending a SIP registration request.
- Send SUBSCRIBE—If checked, the SmartLink will send a SIP *subscribe* to the server specified. This box must be checked for the voicemail message counted and message waiting notification to work.
- SUBSCRIBE Server IP or FQDN—The IP address or fully qualified domain name of the subscription service. If not specified, the SIP subscribes to the SIP server.



Gateway Settings

Dial Plan: []

Figure 13. Gateway Settings section of the SIP window

Gateway Settings

Dial Plan. Refer to appendix C, “Dial plans” on page 80

SIP Extensions

Figure 14. SIP Extensions section of the SIP window

Support PRACK method

Select to enable SIP provisional acknowledgement (PRACK) support as defined in RFC 3262.

Encode SIP URI with user parameter

Select to encode user=phone parameter in SIP URI.

Send INVITE with Timer header

Select to encode Timer header in all INVITE requests for ringing timeout.

Call Hold using C=0.0.0.0

When checked, calls will be held using the call hold method described in RFC 2543. If unchecked, the call hold would follow the RFC 3263 method.

Send NOTIFY

Send out SIP NOTIFY request to transferer for unattended and attended call transfer.

RTP Telephone Event Configuration

This sub-page allows configuration of the out-of-band signaling options for SIP. Select whether OOB telephone event signaling is to be done using the SIP INFO message, or to be done via RFC2833 RTP signaling. For additional information please refer RFC2833.

Figure 15. RTP Telephone Event Configuration section of the SIP window

VoIP VLAN Configuration

This sub-page allows configuration of specific VLAN tags that are to be applied to all SIP signalling and RTP packets used for VoIP calls. These VLAN settings will override any general VLAN settings applied to the interface.

VoIP VLAN Configuration	
Call Signalling Packets	
VLAN Tag:	<input type="text"/>
RTP Packets	
VLAN Tag:	<input type="text"/>

Figure 16. VoIP VLAN Configuration section of the SIP window

SIP Parameters

SIP Parameters			
Hook Flash MIME Type:	<input type="checkbox"/>		
SIP Timer Values (msec)			
SIP T1:	<input type="text" value="500"/>	SIP T2:	<input type="text" value="4000"/>
SIP T4:	<input type="text" value="5000"/>		
RTP Parameters			
RTP Port Min:	<input type="text"/>	RTP Port Max:	<input type="text"/>

Figure 17. SIP Parameters section of the SIP window

Hook Flash MIME Type

This is the MIME Type to be used in a SIP INFO message used to signal hook flash event.

SIP Timer Values (milliseconds)

SIP T1. RFC 3261 T1 value (RTT estimate). Range: 0–64000 milliseconds (default is 5000 msec or 5 seconds).

SIP T2. RFC 3261 T2 value (maximum retransmit interval for non-INVITE requests and INVITE responses). Range: 0–64000 milliseconds (default is 4000 msec or 4 seconds).

SIP T4. RFC 3261 T4 value (maximum duration a message will remain in the network). Range: 0–64 seconds (default is 5000 msec or 5 seconds).

RTP Parameters

RTP Port Min and *RTP Port Max* define a range that contains at least four even-numbered ports (100–106, for example).

NAT Traversal

The screenshot shows a window titled "NAT Traversal". It contains four radio button options: "Outbound Proxy", "Stun Server IP", "UPnP", and "NONE". The "NONE" option is selected. To the right of the "Outbound Proxy" and "Stun Server IP" options are input fields for "Outbound Proxy Port" and "Stun Server Port" respectively. Below the radio buttons is a "Save SIP Settings" button.

Figure 18. NAT Traversal section of the SIP window

Outbound Proxy IP

Type the fully qualified domain name for the outbound proxy server, or type the IP address provided by your service provider.

Outbound Proxy Port

Type the outbound proxy IP port number provided by your service provider.

Stun Server IP

Enter the fully qualified domain name for the stun server, or type the IP address provided by your service provider.

Stun Server Port

Type the stun server port number provided by your service provider.

UPnP

Universal plug-and-play method. This method works with NAT M-ATAs that support UPnP gateway.

NONE

Select this if you will not be using NAT traversal methods.

Saving your work

When you are finished configuring SIP settings, click the **Save SIP Settings** button (see [figure 18](#)) to save the changes.

Audio/CODEC Configuration

Selected	Silence Suppression	Preferred-Codec
<input checked="" type="checkbox"/> G711U	ON	<input type="radio"/>
<input checked="" type="checkbox"/> G711A	ON	<input type="radio"/>
<input type="checkbox"/> G723	ON	<input type="radio"/>
<input type="checkbox"/> G726	ON	<input type="radio"/>
<input type="checkbox"/> G729	ON	<input type="radio"/>

Packetization: 10ms

Jitter Buffer

Adaptive Jitter Buffer: 100ms (maximum playout delay in milliseconds)

Fixed Jitter Buffer: 40ms (fixed playout delay in milliseconds)

Automatically switch to Fixed Jitter Buffer upon fax/modem tone detection

FAX without T.38(Use G.711 fax)

Save CODEC Configuration

Figure 19. Audio/CODEC Configuration window

CODECS

- *Selected* column: Check the codecs that are acceptable to use
- *Silence Suppression* column: Specifies whether silence suppression should be turned on in the SmartLink 4020.
- *Preferred-Codec* column: Select the codec to be used as the first choice when encoding voice

Packetization

Configure the packet sending increment.

Jitter Buffer

Configure the timing of the voice buffering:

- Selection between adaptive or fixed jitter buffer. Default = ADAPTIVE.
- Set the adaptive jitter buffer maximum playout delay. Default = 100ms or Fixed jitter buffer playout delay. Default = 40ms
- Whether or not to automatically switch from an adaptive jitter buffer to a fixed jitter buffer upon fax/modem tone detection

FAX without T.38 (Use G.711 fax)

When checked, T.38 is disabled and the SL4020 will not attempt T.38 session negotiation for fax transmissions initiated by the originator of the fax. The fax transmissions will be transported using G.711 fax pass-through.

The selection of T.38 or G.711 fax pass-through for fax transmissions originated from the SL4020 will be determined by the receiving fax device. T.38 is enabled by default.

To disable T.38, go to *Telephony > CODECS* and select *FAX without T.38 (Use G.711 fax)* (see [figure 19](#) on page 41).

Saving your work

When you are finished configuring CODEC settings, click the **Save CODEC Configuration** button (see [figure 19](#) on page 41) to save the changes.

Phone 1

The *Phone 1* window contains the following sections:

- User Information (see [figure 20](#))
- Supplementary Service Settings (see [figure 21](#) on page 43)
- Dial Out Type (see [figure 22](#) on page 45)
- Call Forward Settings (see [figure 16](#) on page 39)
- Selective Call Forward Settings (see [figure 17](#) on page 39)
- Incoming Call Block (see [figure 18](#) on page 40)
- Distinctive Ring Settings (see [figure 18](#) on page 40)
- HTTP Digest Setting (see [figure 18](#) on page 40)

Note After configuring the sections, click the **Save** button (see [figure 27](#) on page 47) to save the new configuration.

User Information

User Information			
Phone Number	<input type="text" value="jerry"/>	CallerID Name	<input type="text" value="mike"/>
User Name	<input type="text" value="jerry"/>	Password	<input type="text" value="****"/>
Port	<input type="text" value="5060"/>	SIP Registration status	Registered
Supplementary Services Activation			
Service	Enable?	Service	Enable?
Call Forward All	<input type="text" value="No"/>	Call Forward on Busy	<input type="text" value="No"/>
Selective Call Forward	<input type="text" value="No"/>	Conditional Call Forward	<input type="text" value="No"/>
Three Way Conferencing	<input type="text" value="Yes"/>	Call Waiting	<input type="text" value="Yes"/>
Incoming Call Block	<input type="text" value="Yes"/>	Anonymous Call Reject	<input type="text" value="No"/>
Distinctive Ring	<input type="text" value="No"/>	Caller ID	<input type="text" value="Yes"/>
Call Transfer	<input type="text" value="Yes"/>	Call Return	<input type="text" value="Yes"/>
Dialing by IP Address	<input type="text" value="No"/>	Do Not Disturb	<input type="text" value="No"/>
Speed Dial	<input type="text" value="No"/>	Self Caller ID Block	<input type="text" value="No"/>
Message Waiting Indicator	<input type="text" value="Yes"/>	Outgoing Call Block	<input type="text" value="Yes"/>

Figure 20. User Information section of Phone 1 window

Phone Number

Enter the telephone number or the user part of the SIP registration.

User Name

Enter the user name that will be used for validation of the VoIP SIP registration or call invitation.

Port

Specify the signaling port.

CallerID Name

Enter the caller ID name.

Password

Enter the password.

Supplementary Service Activation

These settings enable or disable each of following calling features. Most features can also be enabled or disabled by using the telephone handset (see section “Supplementary Service Keys” on page 58 for details).

Supplementary Services Activation			
Service	Enable?	Service	Enable?
Call Forward All	No	Call Forward on Busy	No
Selective Call Forward	No	Conditional Call Forward	No
Three Way Conferencing	Yes	Call Waiting	Yes
Incoming Call Block	Yes	Anonymous Call Reject	No
Distinctive Ring	No	Caller ID	Yes
Call Transfer	Yes	Call Return	Yes
Dialing by IP Address	No	Do Not Disturb	No
Speed Dial	No	Self Caller ID Block	No
Message Waiting Indicator	Yes	Outgoing Call Block	Yes

Figure 21. Supplementary Service Settings section of Phone 1 window

Call Forward All

Enable call forward all service—All received calls will be forwarded to the destination specified under the call forwarding settings.

Selective Call Forward

Enable call forward no answer service—All received calls that are not answered will be forwarded to the destination specified under the call forwarding settings.

Three-Way Conferencing

Enable three way conference service—This service enables you to add a third party to an existing two-way conversation, and hold a three-party conference call.

Incoming Call Block

Enable incoming call block service—Allows for selected inbound caller IDs to be blocked.

Distinctive Ring

Enable distinctive ringing service—This service allows additional telephone numbers to be added to an existing telephone line and when a caller dials one of these “distinctive ringing” numbers, the telephone will ring in a unique pattern to indicate which number is being dialed.

Call Transfer

Enable call transfer service—This service allows you to transfer calls to another number.

Dialing by IP Address

Enable IP dialing service—This service allows user IP addresses to be used to make calls.

Speed Dial

Enable speed dial service.

Message Waiting Indicator

Enable MWI service—The message-waiting indicator (MWI) is a common feature of telephone networks and uses an audible indication (such as a special dial tone) to indicate that a voice mail message is waiting.

Call Forward on Busy

Enable call forward on busy service.

Conditional Call Forward

Enable call forward selective service.

Call Waiting

Enable call waiting service.

Anonymous Call Reject

Enable block anonymous calls service—When enabled, calls from anonymous callers will be blocked.

Caller ID

Enable caller ID service.

Call Return

Enable call return service—When enabled, allows you to return a call to the last incoming call, whether the call was answered or not.

Do Not Disturb

Enable do not disturb service.

Self Caller ID Block

Enable blocking self caller ID shown in the outgoing message.

Outgoing Call Block

Enable blocking of outgoing calls based on the digit pattern specified under outgoing call block.

Dial Out Type



Dial Out Type	
Dial Out Type:	NORMAL
Hot Line Number:	
Warm Line Number:	227

Figure 22. Dial Out Type section of Phone 1 window

Dial Out Type

Enable Hot-Line and Warm-Line services. To achieve this, one sequence in the dial plan must start with a pause, with a 0 delay for a Hot Line, and a non-zero delay for a Warm Line.

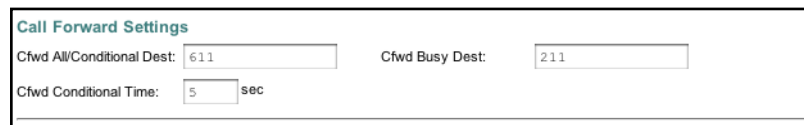
Hot Line Number

Input the number for Hot Line function—This number will be called immediately when the telephone goes off-hook.

Warm Line Number

Input the number for Warm Line function—The warm line function provides a delay period after the telephone goes off-hook for the user to dial a number different than that specified for the warm line. If the delay period has expired with no number being dialed, the warm line number will be dialed. The delay period is set under *System > Localization*.

Call Forward Settings



Call Forward Settings	
Cfwd All/Conditional Dest:	611
Cfwd Busy Dest:	211
Cfwd Conditional Time:	5 sec

Figure 23. Call Forward Settings section of Phone 1 window

Cfwd All Dest

Input the destination for all call forwarding.

Cfwd Busy Dest

Input the destination for all busy call forwarding.

Selective Call Forward Settings

Selective Call Forward Settings			
Incoming caller #1	<input type="text" value="lukan"/>	forward destination #1	<input type="text" value="482"/>
Incoming caller #2	<input type="text" value="joe@yourcompany.com"/>	forward destination #2	<input type="text" value="413"/>
Incoming caller #3	<input type="text"/>	forward destination #3	<input type="text"/>
Incoming caller #4	<input type="text"/>	forward destination #4	<input type="text"/>
Incoming caller #5	<input type="text"/>	forward destination #4	<input type="text"/>
Incoming caller #6	<input type="text"/>	forward destination #4	<input type="text"/>
Incoming caller #7	<input type="text"/>	forward destination #7	<input type="text"/>
Incoming caller #8	<input type="text"/>	forward destination #8	<input type="text"/>

Figure 24. Selective Call Forward Settings section of Phone 1 window

Incoming caller #1–8

Up to 8 incoming calls can be selected for call forwarding.

Forward destination #1–8

Up to 8 destinations to which incoming calls can be forwarded.

Incoming Call Block

Incoming Call Block	
Block Caller ID:	<input type="text" value="joe@yourcompany.com"/>

Figure 25. Incoming Call Block section of Phone 1 window

Block Caller ID

Specify a Caller ID for call block.

Outgoing Call Block

Outgoing call block allows for the blocking of an outbound call based on the digit pattern dialed. For example: outbound call blocking may be used to prevent an end user from dialing long distance or toll calls. Outbound call blocking uses the same digit matching process as the dial plan described in appendix C, “Dial plans” on page 80.

Outgoing Call Block	
Digit Pattern	<input type="text"/>

Figure 26. Incoming Call Block section of Phone 1 window

Digit Pattern

The *Digit Pattern* box is where you can specify the outgoing call block pattern.

Example: Specifying an outgoing call block pattern of *002|009|0204* will block all outbound calls to numbers 002, 009, or 0204.

HTTP Digest Setting

SIP INVITE must contain a valid Authorization header that is based on an Auth ID and a password using MD5 digest algorithm. The Auth ID must be specified in the username parameter in the Authorization header.

Figure 27. HTTP Digest Setting section of Phone 1 window

Saving your work

When you are finished configuring settings, click the **Save** button (see [figure 27](#)) to save the changes.

Speed Dial

Figure 28. Speed Dial window

Line 1 Speed Dial Settings

Speed Dial Serv

Enable Speed Dial Service.

Speed Dial 1–8 Phone Number/IP Dialing

Target 1–8 phone number (or URL) assigned to speed dial.

Saving your work

When you are finished configuring settings, click the **Save Settings** button (see [figure 28](#)) to save the changes.

Chapter 6 **System**

Chapter contents

Set Security Password	51
Web Page Protect	51
New Root Password	51
New User Password	51
Confirm new password	51
Saving your work	51
Configuration	52
Syslog Service	52
Enable Syslog	52
Syslog Server	52
Debug Service	53
Enable Debug	53
Debug Server	53
Debug Connection Port	53
NTP Service	53
NTP Server	54
Time Zone	54
Adjust clock for daylight savings	54
Saving your work	54
Service Access	54
Enable HTTP	54
Enable Ping Reply	54
Saving your work	54
Localization	55
Call Progress Tones	55
Dial Tone	56
Prompt Tone	56
Confirm Tone	56
Holding Tone	56
Busy Tone	56
Ring Back Tone	56
Off Hook Warning	56
Distinctive Ring Settings	57
Supplementary Service Keys	58
Call forward All	58
Call forward on Busy	58
Call forward selective	58
Call Waiting	58
Incoming Call Block	58

Block Anonymous Calls	58
Distinctive Ring	58
Warm Line	58
Do Not Disturb	59
IP Dialing	59
Speed Dialing	59
Income Caller ID	59
Self Caller ID Block	59
Calling Prefix Keys	59
Call Return	59
Warm Line	59
Speed Dial	60
Call Forward All	60
IP Dialing	60
Call Hold	60
Call Waiting (call alternative)	60
Call Conference	60
Call Conference Drop	60
Call Transfer	60
Voicemail access	61
Call Forward Busy Destination	61
Hotline	61
Control Timer Values	61
Hook Flash Timer	61
SIP Session Timer value	61
Conditional Call Forwarding Timer	62
Warm Line Delay	62
Interdigit Timer	62
Offhook Idle Time	62
Offhook Warning tone time	62
FXS Port	62
FXS port Input Gain	62
FXS port Output Gain	62
Caller ID Method	62
“Call Progress Tones” Synchronization	62
Ring Setting	63
Ring Waveform	63
Ring Frequency	63
Ring Voltage	63
FXS Port Polarity Configuration	63
Idle Polarity	63
Caller Conn Polarity	63
Saving your work	63
SNMP Configuration.....	64

SNMP Trap Configuration	64
IP address	64
Trap Community	64
SNMP Community Configuration	64
Read Community	64
Write Community	64
SNMP System Configuration	64
System Description	64
System Object Id	64
Saving your work	64
Auto Upgrade.....	65
Routine Upgrade every xxx day(s)	65
Enable Auto Upgrade	65
Auto Upgrade Protocol	65
Upgrade Server	65
Auto Upgrade URL	65
Saving your work	65
Manual Upgrade.....	66
Reload	67

Set Security Password



The screenshot shows a web configuration interface with a blue sidebar on the left containing a tree view of menu items: Home, Network - LAN (with sub-items Status, Settings, ToS), Telephony (with sub-items VoIP Status, SIP, CODECS, Phone 1, Speed Dial), System (with sub-items Security, Configuration), and Configuration. The main content area is titled "Set Security Password" and contains the following elements: a status message "Password is currently installed", a "Web Page Protect" dropdown menu set to "NO", and four password input fields: "New Root Password:", "Confirm new password:", "New User Password:", and "Confirm new password:". A "Done" button is located at the bottom left of the main content area.

Figure 29. Set Security Password window

Two levels of system configuration are available: user level and system level. Access to each level is password controlled.

Web Page Protect

Enable or disable web access protection. When set to *YES*, a password will be required to access the SmartLink M-ATA web configuration interface.

New Root Password

Type the administrator's password.

New User Password

Type the user's password.

Confirm new password

Re-enter the password for confirmation.

Saving your work

When you are finished configuring security settings, click the **Save Settings** button to save the changes.

Configuration

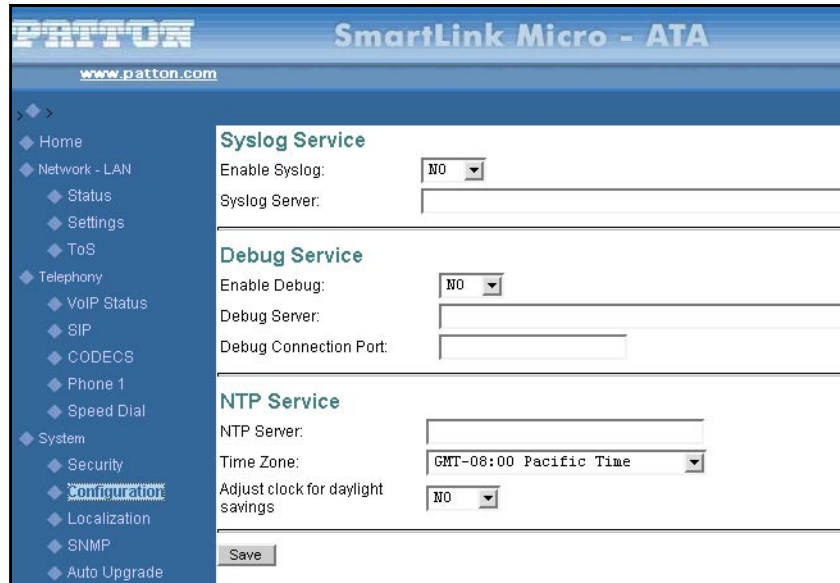


Figure 30. Configuration window

Syslog Service

Enable Syslog

Enable or disable system logging. During normal operations, Syslog should be turned off. A sample Syslog record of a complete call is shown below.

SYSLOG

```

11-09-2005    11:28:17    Local0.Info    209.49.110.185    Nov 09
08:28:34 syslog: [1234] On-Hook [7190]
11-09-2005    11:28:06    Local0.Info    209.49.110.185    Nov 09
08:28:23 syslog: [1234] Call-Out [7190]
11-09-2005    11:27:50    Local0.Info    209.49.110.185    Nov 09
08:28:07 syslog: [1234] Call-Out [190]
11-09-2005    11:16:32    Local0.Info    209.49.110.185    Nov 09
08:16:49 syslog: [1234] On-Hook [555]
11-09-2005    11:16:16    Local0.Info    209.49.110.185    Nov 09
08:16:33 syslog: [1234] Call-Out [7190]
11-09-2005    11:15:38    Mail.Emerg     209.49.110.185    Jan 01
00:00:00 syslog: DownloadConfig:No, DownloadImage:No
11-09-2005    11:15:36    Local0.Info    209.49.110.185    Jan 01
00:00:00 syslog: Check Autoupgrade

```

Syslog Server

Specify the syslog server IP address or DNS name. This feature specifies the server for logging SmartLink M-ATA system information and critical events.

Debug Service

Enable Debug

Enable or disable System Debug.

Debug Server

The debug server IP address and port. This specifies the server for logging SmartLink M-ATA debug information. Debug information can be sent to a syslog server. A sample record of debug output is shown below:

SYSLOG

```
07-07-2005 14:49:09 Local7.Debug 10.10.22.13 **EndPoint State: StateConnect ==> StateOnHook
in src/common/endpoint.c: 333
07-07-2005 14:49:09 Local0.Info 10.10.22.13 Jul 07 11:49:09 syslog: [usersjphone] On-Hook
[100]
07-07-2005 14:49:03 Local7.Debug 10.10.22.13 !!!! Handset 349b4 received event 52 in file src/
common/fxs/ConnectFxs.c line 466
07-07-2005 14:48:59 Local7.Debug 10.10.22.13 The EnvetAddress is f and length is 1.
07-07-2005 14:48:59 Local7.Debug 10.10.22.13 **EndPoint State: StateWaiting ==> StateConnect
in src/common/fxs/WaitingFxs.c: 116
07-07-2005 14:48:59 Local7.Debug 10.10.22.13 call 468bc ==>CallChangeState():: from
CALL_ALERT to CALL_CONNECT in file src/common/fxs/WaitingFxs.c line 114
07-07-2005 14:48:59 Local7.Debug 10.10.22.13 APP rcv rsp 200 of req 32769
07-07-2005 14:48:59 Local7.Debug 10.10.22.13 call 468bc ==>CallChangeState():: from
CALL_OUTGOING to CALL_ALERT in file src/common/fxs/WaitingFxs.c line 58
07-07-2005 14:48:59 Local7.Debug 10.10.22.13 APP rcv rsp 180 of req 32769
07-07-2005 14:48:58 Local7.Debug 10.10.22.13 !!!! Handset 349b4 received event 26 in file src/
common/fxs/WaitingFxs.c line 212
07-07-2005 14:48:58 Local7.Debug 10.10.22.13 APP rcv rsp 100 of req 32769
07-07-2005 14:48:58 Local7.Debug 10.10.22.13 **EndPoint State: StateOffHook ==> StateWaiting
in src/common/fxs/OffHookFxs.c: 418
07-07-2005 14:48:58 Local7.Debug 10.10.22.13 Dial numbers is 100
07-07-2005 14:48:58 Local0.Info 10.10.22.13 Jul 07 11:48:58 syslog: [usersjphone] Call-Out
[100]
07-07-2005 14:48:58 Local7.Debug 10.10.22.13 call 0 ==>CallChangeState():: from CALL_IDLE to
CALL_OUTGOING in file src/common/call.c line 80
07-07-2005 14:48:58 Local7.Debug 10.10.22.13 Codec[2]=8
07-07-2005 14:48:58 Local7.Debug 10.10.22.13 Codec[1]=18
07-07-2005 14:48:58 Local7.Debug 10.10.22.13 Codec[0]=0
07-07-2005 14:48:54 Local7.Debug 10.10.22.13 The EnvetAddress is 100 and length is 3.
07-07-2005 14:48:53 Local7.Debug 10.10.22.13 The EnvetAddress is 10 and length is 2.
07-07-2005 14:48:53 Local7.Debug 10.10.22.13 The EnvetAddress is 1 and length is 1.
07-07-2005 14:48:51 Local7.Debug 10.10.22.13 **EndPoint State: StateOnHook ==> StateOffHook
in src/common/fxs/OnHookFxs.c: 55
```

Debug Connection Port

The port number of the debug server to be used for receiving debug messages from the SmartLink M-ATA. Use port 412 to send debug output to a syslog server.

NTP Service

The network time protocol (NTP) synchronizes timekeeping among a set of distributed time servers and clients. An NTP client synchronizes the local clock with some other time source, usually an NTP server. A list of public NTP servers is available at <http://ntp.isc.org/bin/view/Servers/WebHome>.

NTP Server

Specify the IP address of the NTP server.

Time Zone

Select the GMT time zone for your location. For help in determining the time zone, go to the *UTC/GMT Conversion* website at <http://www.dxing.com/utcgmt.htm>.

Adjust clock for daylight savings

Select *Yes* if you want the M-ATA to automatically compensate for daylight savings time.

Saving your work

When you are finished configuring settings, click the **Save** button to save the changes.

Service Access

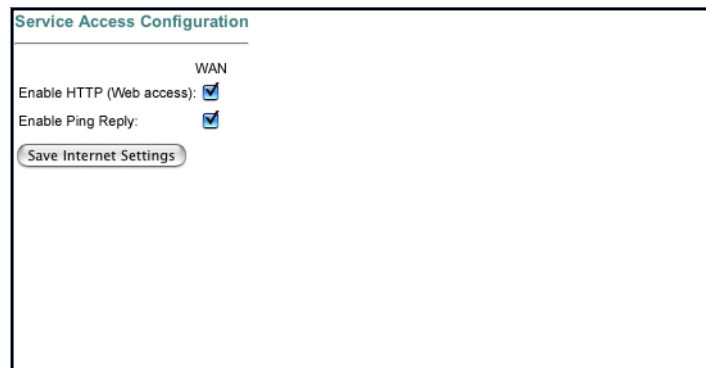


Figure 31. Service Access window

Enable HTTP

When checked access to the web configuration interface (HTTP) of the SmartLink M-ATA through the WAN interface port is enabled.

Note If the *Enable HTTP* option is not checked, your WAN service provider will no longer be able to access the SmartLink M-ATA to make configuration changes.

Enable Ping Reply

When checked, the SmartLink M-ATA will reply to ping requests received on the WAN port. If the option is not checked, the SmartLink M-ATA will not respond to ping requests received on the WAN port.

Saving your work

When you are finished configuring settings, click the **Save Service Access Settings** button to save the changes.

Localization

The *Localization* window contains the following sections:

- Call Progress Tones (see [figure 32](#) on page 56)
- Distinctive Ring Setting (see [figure 33](#) on page 57)
- Supplementary Service Keys (see [figure 34](#) on page 58)
- Control Timer Values (see [figure 36](#) on page 61)
- FXS Port (see [figure 37](#) on page 62)
- Ring Setting (see [figure 38](#) on page 63)
- FXS Port Polarity (see [figure 39](#) on page 63)

Note After configuring the sections, click the **Save** button (see [figure 39](#) on page 63) to save the new configuration.

Call Progress Tones

The dialtone, confirmation, busy, ringback and offhook warning call progress tones that are played to the handset can be set manually or the tones will be set automatically to the country specified under *Call ID method* if the checkbox, *Call Progress Tones synchronization* has been selected.

To set the call progress tones manually specify:

```
[frequency 1]@[Energy 1]+[frequency 2]@[Energy 2]#ON([ms]),OFF([ms]),R
```

Where:

- *Frequency 1* and *frequency 2* are the frequency of the tone to be played.
- *Energy 1* and *energy 2* are the energy level of the tone to be played in dBm.
- *ON(ms)* is the duration that the tone will be played in milliseconds.
- *OFF(ms)* is the duration of the pause between cycles of playing the tone.
- *R* indicates the tone should be repeated using the ON and OFF durations specified

To specify a 350-Hz and 480-Hz tone that will be played at 15 dB for one second and then pause for 0.5 seconds, you would specify:

```
350@-15+480@-15#ON(1000),OFF(500),R
```

The following example would repeat playing of a tone of 280 Hz at -8 dBm for 0.3 seconds with a pause of 0.1 second and a tone of 550 Hz for 0.8 seconds with a pause of 0.15 seconds.

```
280@-8+550@-20#ON(300),OFF(100),ON(800),OFF(150),R
```

Localization	
Call Progress Tones	
DIALTONE	3300-13+4400-13#ON(1000),R
CONFIRM	3500-13+4400-13#[ON(100),OFF(100)]3,OFF(1000),R
BUSY	4400-13#ON(500),OFF(500),R
RINGBACK	4400-13#ON(1500),OFF(3500),R
OFF HOOK WARNING	14000-3+20600-3+24500-3+26000-3#ON(100),OFF(100),R

Figure 32. Call Progress Tones section of Localization window

Dial Tone

Played when prompting the user to enter a phone number.

Prompt Tone

Played when prompting the user to enter a call forward phone number.

Confirm Tone

This should be a brief tone to notify the user that the last input value has been accepted.

Holding Tone

Indicate to the local user that the far end has placed the call on hold.

Busy Tone

Played when a 486 RSC is received for an outbound call.

Ring Back Tone

Played for an outbound call when the far end is ringing.

Off Hook Warning

Played when the subscriber does not place the handset on the cradle properly.

Distinctive Ring Settings

The SmartLink M-ATA Distinctive Ring Settings window (see [figure 33](#)) enables you to specify up to 8 sets of distinctive ring cadences.

Figure 33. Distinctive Ring Setting section of Localization window

The following is a sample ring cadence pattern configuration:

```
Timeval ::= time in milliseconds
Repeatval ::= # of cycles to repeat
Tonename ::= "RING_0" | "RING_1" | "RING_2" | "RING_3" | "RING_4"
| "RING_5" | "RING_6" | "RING_7" | "RING_8" | "RING_9"
Idle ::= "IDLE"
Active ::= "ON" | "OFF"
Inactive ::= Idle "(" Timeval ")"
Active ::= Active "(" Timeval ")"
Sequence ::= Active | Active "," Sequence
Repetition ::= "[" Sequence "]" Repeatval
Repeat ::= "R"
Cycle ::= Sequence | Repetition
Fullsequence ::= Cycle | Cycle "," Fullsequence
Cadence ::= Fullsequence | Fullsequence "," Repeat | Fullsequence
"," Inactive "," Repeat
Ring ::= Cadence
```

Note The Bellcore standard ring cadence patterns are shown in [table 2](#).

Table 2. Bellcore standard ring cadence patterns

Name	Value
RING_0	ON(2000), IDLE(4000), R
RING_1	ON(800), OFF(400), ON(800), IDLE(4000), R
RING_2	ON(400),OFF(200)]2,ON(800),IDLE(4000),R
RING_3	ON(300), OFF(200), ON(1000), OFF(200), ON(300), IDLE(4000), R
RING_4	ON(500)
RING_5	

Supplementary Service Keys

Supplementary Service Keys are key sequences that can enable and disable certain calling features from an analog phone handset attached to the SmartLink M-ATA.

Note The supplementary service keys must begin with a ‘*’, ‘#’ character, or f (flash hook) and follow a 1 or 2 numeric digit(s).

Supplementary Service Keys		
	Enable Sequence	Disable Sequence
Call forward All	*97	#97
Call forward on Busy	*98	#98
Call forward selective	*96	#96
Call Waiting	*91	#91
Incoming Call Block	*95	#95
Block Anonymous Calls	*94	#94
Distinct Ring	*90	#90
Warm Line	*99	#99
Do Not Disturb	*82	#82
IP Dialing	*80	#80
Speed Dialing	*81	#81
Incoming Caller ID	*92	#92
Self Caller ID Block	*93	#93

Figure 34. Supplementary Service Keys section of Localization Window

Call forward All

Enables or disables the unconditional call forwarding feature.

Call forward on Busy

Enables or disables the call forwarding on busy feature.

Call forward selective

Enables or disables the call forwarding on busy feature.

Call Waiting

Enables or disables the call waiting feature.

Incoming Call Block

Enables or disables the blocking of incoming calls based on the phone number of the caller feature.

Block Anonymous Calls

Enables or disables the blocking of incoming calls that are using the anonymous caller ID feature.

Distinctive Ring

Enables or disables the distinctive ring tones based on the phone number of the calling party feature.

Warm Line

Enables or disables the warm line calling feature.

Do Not Disturb

Enables or disables the do not disturb feature.

IP Dialing

Enables or disables dialing by IP address.

Speed Dialing

Enables or disables the speed dialing feature.

Income Caller ID

Enables or disables the display of incoming caller ID feature.

Self Caller ID Block

Enables or disables the blocking of transmission of caller ID feature for outgoing calls.

Calling Prefix Keys

Calling Prefix Keys and other configuration parameters define the behavior of the calling feature.

Note Following function key must be start by ‘*’, ‘#’ character, or f (flash hook) and follow a 1 or 2 numeric digit(s).

Calling Prefix Keys			
Call Return	*60	Warm Line	*70
Speed Dial	*71	IP Dialing	*72
Call Forward All Destination	*73	Call Forward Busy Destination	*74
Hotline	*75	Call Hold	f1
Call Alternative	f*	Call Conference	f7
Call Conference drop	f8	Call Transfer	f4
VoiceMail dial	*86		
[Note] (f)=Flash hook			

Figure 35. Calling Prefix Keys section of Localization Window

Call Return

Automatically places a call to the number of the last call received.

Warm Line

Configures the number to call automatically after a delay period once the telephone goes off hook. To configure the warm line number enter the calling prefix key (*70 is the factory default), the number to dial and then on hook the phone. You will hear a second dial tone after entering the calling prefix key. Before using warm-line calling it must be enabled using the activate sequence defined in the supplementary service table.

Example:

*7013019751000 and on-hook the phone. When the handset is lifted, 301951000 will be called if another number is not entered within the delay period specified under *System > Localization*.

Speed Dial

Configures speed dialing keys. To configure speed dialing numbers enter the calling prefix key (*71 is the factory default), the speed dial key, the number to dial and then on hook the phone. You will hear a second dial tone after entering the calling prefix key. Speed dial calling must be enabled using the activate sequence defined in the supplementary service table.

Example:

*7123019751000 and on-hook the phone. When 2 is pressed on the telephone key pad, 3019751000 will be called.

Call Forward All

Configures the number to unconditionally forward all calls to. To configure call forward all enter the calling prefix key (*73 is the factory default), the number to dial and then on hook the phone. You will hear a second dial tone after entering the calling prefix key. Before using, *Call Forward All* must be enabled using the activate sequence defined in the supplementary service table.

Example:

*733019751000 and on-hook the phone. All calls received will forward to 3019751000.

IP Dialing

Allows direct calling using an IP address. To use calling by IP address enter the calling prefix key (*72 is the factory default), the * key, the IP address using the * key to between octets of the IP address, the * key and the port number. You will hear a second dial tone after entering the calling prefix key. Before using direct IP address dialing must be enabled using the activate sequence defined in the supplementary service table.

Example:

To place a call to IP address 192.168.1.20:5061 enter *73*192*168*1*20*5060.

Call Hold

Configures the key sequence to place a call on call on hold – The factory default is flash hook 1.

Call Waiting (call alternative)

Configures the key sequence to switch between calls. The factory default is flash hook *.

Call Conference

Configures the key sequence to conference two calls together. The factory default is flash hook 7.

Call Conference Drop

Configures the key sequence to drop the last call that was added to the conference the conference. The factory default is flash hook 8.

Call Transfer

Configures the key sequence for call transfer. The factory default is flash hook 4.

Voicemail access

Configures the key sequence for call for voicemail access. The factory default is flash hook *86. If the voicemail key sequence is entered on the phone a call will automatically be placed to the configured service provider using the configured authentication information.

Call Forward Busy Destination

Configures the number to forward calls to when busy. To configure call forward busy enter the calling prefix key (*74 is the factory default), the number to dial and then on hook the phone. You will hear a second dial tone after entering the calling prefix key. Before using, call forward busy must be enabled using the activate sequence defined in the supplementary service table.

Example:

*743019751000 and on-hook the phone. All calls received will forward to 3019751000 when the phone is busy.

Hotline

Configures the number to call when the telephone goes off hook. To configure the hotline number enter the calling prefix key (*75 is the factory default), the number to dial and then on hook the phone. You will hear a second dial tone after entering the calling prefix key. Before using hotline calling it must be enabled using the activate sequence defined in the supplementary service table.

Example:

*753019751000 and on-hook the phone. When the handset is lifted, 3019751000 will immediately be called.

Control Timer Values

Control Timer Values			
Hook Flash Timer: (100 - 1100 ms)	<input type="text" value="1100"/>	ms	SIP Session Timer value: <input type="text"/>
Conditional Call Forwarding Timer:	<input type="text" value="10"/>	sec	Warm Line Delay: <input type="text" value="6000"/>
Interdigit Timer:	<input type="text" value="4000"/>	ms	Offhook Idle Time: <input type="text" value="8000"/>
Offhook Warning Tone Time:	<input type="text" value="12000"/>	ms	

Figure 36. Control Timer Values section of Localization window

Hook Flash Timer

Maximum on-hook time before off-hook to qualify as hookflash. More than this value and the on-hook event is treated as on-hook (hanging up the call).

Minimum on-hook time before off-hook to qualify as hookflash. At less than this value, the on-hook event is ignored.

SIP Session Timer value

The amount of time the SmartLink M-ATA will wait during an active call to send repeated re-invites on active calls to allow the SIP server to determine the status of a call.

Conditional Call Forwarding Timer

Specified a time period as a call forward condition. After the number of seconds specified, the conditional call forwarding process will be performed.

Warm Line Delay

Specify a time period as a delay time for warm line dialing. If warm line is enabled, then the SmartLink M-ATA will wait this amount of time after the handset is off-hook before dialing the warm line phone number.

Interdigit Timer

The number of seconds the SmartLink M-ATA will wait for the caller to input a subsequent digit of the dialed number. If the timer value is exceeded before the dial plan is matched (see section “Gateway Settings” on page 37), the busy tone will be played to the caller.

Offhook Idle Time

If the handset is off-hook with no dialing activity for longer than the time specified, then the busy tone will be played. The default is 6000 ms (6 seconds).

Offhook Warning tone time

If the handset is off-hook with no dialing activity for longer than the time specified, then the off-hook warning tone will be played. The default is 12000 ms (12 seconds). This value should be greater than the off-hook idle time (see section “Offhook Idle Time”).

FXS Port

Choose the correct country for a proper impedance match.

The screenshot shows the 'FXS Port' configuration section. It includes the following fields and options:

- FXS Port Input Gain:** A text input field with the unit 'db (-12 ~ 18)'.
- FXS Port Output Gain:** A text input field with the unit 'db (-12 ~ 18)'.
- Caller ID Method:** A dropdown menu currently showing 'France'.
- Call Progress Tones Synchronization:** A checkbox that is checked.

Figure 37. FXS Port Polarity Configuration section of Localization window

FXS port Input Gain

Adjust the input gain level for FXS port.

FXS port Output Gain

Adjust the output gain level for FXS port.

Caller ID Method

Specifies the country-specific Caller ID format.

“Call Progress Tones” Synchronization

When “Call Progress Tones” Synchronization is selected, the Caller ID presentation method value is used to automatically set the county-specific call progress tones.

Ring Setting



Ring Setting
Ring Waveform: Ring Frequency:
Ring Voltage:

Figure 38. Ring Setting section of Localization window

Ring Waveform

Specify the ring tone waveform.

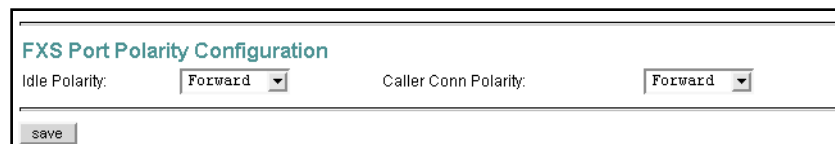
Ring Frequency

Specify the ring tone frequency.

Ring Voltage

Specify the ring tone voltage.

FXS Port Polarity Configuration



FXS Port Polarity Configuration
Idle Polarity: Caller Conn Polarity:

Figure 39. FXS Port Polarity section of Localization window

Idle Polarity

Polarity before call connected.

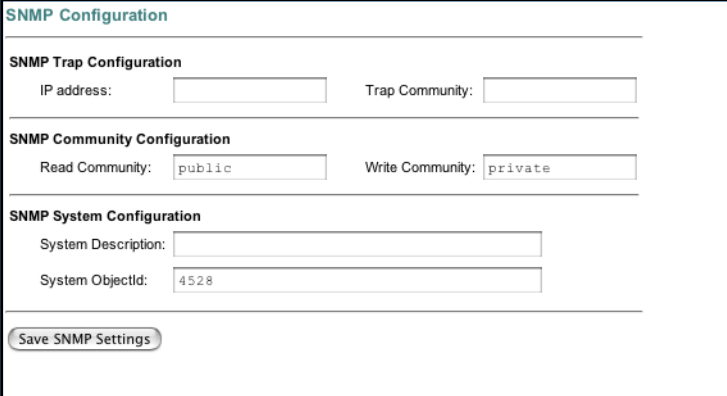
Caller Conn Polarity

Polarity after outbound call connected.

Saving your work

When you are finished configuring settings, click the **Save** button (see [figure 39](#)) to save the changes.

SNMP Configuration



The screenshot shows the 'SNMP Configuration' window. It is divided into three sections: 'SNMP Trap Configuration', 'SNMP Community Configuration', and 'SNMP System Configuration'. The 'SNMP Trap Configuration' section has two input fields: 'IP address' and 'Trap Community'. The 'SNMP Community Configuration' section has two input fields: 'Read Community' (with the value 'public') and 'Write Community' (with the value 'private'). The 'SNMP System Configuration' section has two input fields: 'System Description' and 'System Objectid' (with the value '4528'). At the bottom left of the window is a button labeled 'Save SNMP Settings'.

Figure 40. SNMP Configuration window

SNMP Trap Configuration

IP address

Trap host IP address.

Trap Community

The community name used by the SNMP manager to verify traps. The default value is *public*.

SNMP Community Configuration

Read Community

The community name used by the SNMP manager when reading SNMP data items from a client MIB. The default value is *public*.

Write Community

The community name used by the SNMP manager when setting SNMP data items in a client's MIB. The default value is *public*.

SNMP System Configuration

System Description

Description of the unit (e.g. "John's phone")

System Object Id

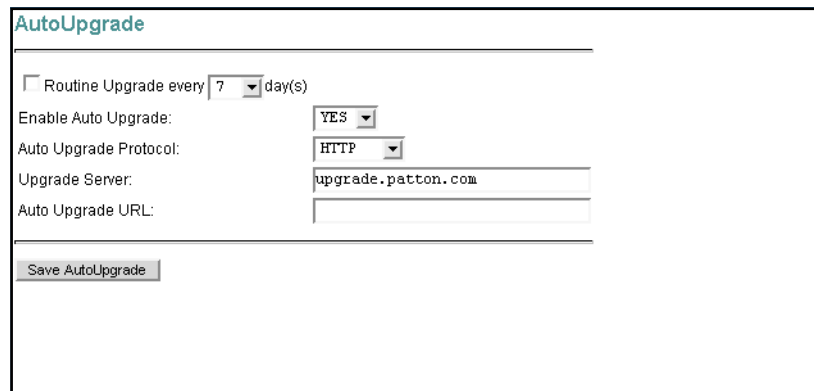
A vendor's enterprise ID

Saving your work

When you are finished configuring settings, click the **Save SNMP Settings** button to save the changes.

Auto Upgrade

The SmartLink M-ATA family includes a configuration and firmware download manager server that allows for the updating of large numbers of SmartLink M-ATAs from a central location. By factory default, all SmartLink M-ATA units are set with auto-update on and to access the Patton auto-update server. Many of Patton's carrier customers have chosen to setup their own auto-update server to provide service specific information for their end users. Patton recommends that end users do not change the auto-update server or set "Enable auto-upgrade" to "NO" without consulting with their service provider.



The screenshot shows a web-based configuration window titled "AutoUpgrade". At the top, there is a checkbox labeled "Routine Upgrade every 7 day(s)". Below this, there are four rows of configuration options: "Enable Auto Upgrade:" with a dropdown menu set to "YES"; "Auto Upgrade Protocol:" with a dropdown menu set to "HTTP"; "Upgrade Server:" with a text input field containing "upgrade.patton.com"; and "Auto Upgrade URL:" with an empty text input field. At the bottom of the window, there is a button labeled "Save AutoUpgrade".

Figure 41. AutoUpgrade window

Routine Upgrade every xxx day(s)

If selected, the SmartLink M-ATA will check for updates at system reload time and on a periodic basis based on the number of days selected.

Enable Auto Upgrade

Enable or disable auto upgrade—If enabled, the SmartLink M-ATA will automatically check the upgrade server for new system firmware and software upon reload, power cycle, or when the routine upgrade interval expires.

Auto Upgrade Protocol

Select the protocol for auto upgrade

Upgrade Server

Specify the auto upgrade server IP address

Auto Upgrade URL

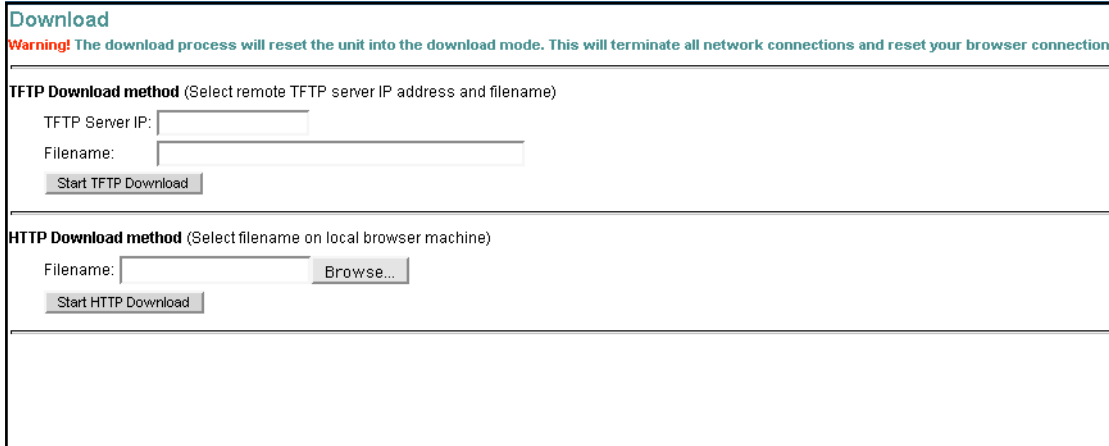
Specify the auto upgrade server by URL. This field is dependent on the auto upgrade service package installation. The default value is `iadmgr`.

Saving your work

When you are finished configuring settings, click the **Save AutoUpgrade** button to save the changes.

Manual Upgrade

For both **HTTP** and **TFTP** methods, the device will reboot itself into the downloader mode if the main application is executing, and proceed with the ROM file download and permanent write of the application to the device's flash memory. During download of new firmware images, the LEDs on the SmartLink M-ATA will flash sequentially. Typically, HTTP downloads take about 5–10 minutes. After the download is completed, the download status page will be displayed.



The screenshot shows a web interface titled "Download". At the top, there is a warning message: "Warning! The download process will reset the unit into the download mode. This will terminate all network connections and reset your browser connection." Below the warning, there are two sections for selecting a download method. The first section is "TFTP Download method (Select remote TFTP server IP address and filename)", which includes input fields for "TFTP Server IP:" and "Filename:", and a "Start TFTP Download" button. The second section is "HTTP Download method (Select filename on local browser machine)", which includes a "Filename:" input field with a "Browse..." button next to it, and a "Start HTTP Download" button.

Figure 42. Manual Upgrade window

Reload



Figure 43. Reload window



Reloading the system will terminate all network connections and restart your browser connection.

Chose the *Reload and execute Main Application* option, for execution of the main application which you have configure, once you reload the system.

Chose the *Reload and execute Downloader Application* option, to being downloading, once you reload the system.

Chapter 7 **Documentation**

Chapter contents

Introduction.....	69
-------------------	----

Introduction

Clicking the **Documentation** link (see [figure 44](#)) connects to the Patton website to display the most current version of the *SmartLink M-ATA Getting Started Guide* in portable document format (PDF).



Figure 44. Documentation link

Chapter 8 **Logout**

Chapter contents

Introduction.....	71
-------------------	----

Introduction

Clicking *Logout* (see [figure 45](#)) displays *Logout—Sure to Logout?*.

If you want to exit from the SmartLink management utility, click the **Logout** button (see [figure 45](#)).



Figure 45. Logout window

You will be returned to the password verification page (see [figure 46](#)).

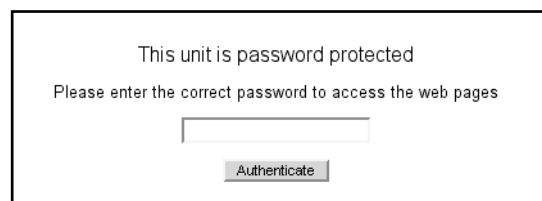


Figure 46. Password verification page

Chapter 9 **Contacting Patton for assistance**

Chapter contents

Introduction	73
Contact information	73
Patton support headquarters in the USA	73
Alternate Patton support for Europe, Middle East, and Africa (EMEA)	73
Warranty Service and Returned Merchandise Authorizations (RMAs)	73
Warranty coverage	74
Out-of-warranty service	74
Returns for credit	74
Return for credit policy	74
RMA numbers	74
Shipping instructions	74

Introduction

This chapter contains the following information:

- “Contact information”—describes how to contact Patton technical support for assistance.
- “Warranty Service and Returned Merchandise Authorizations (RMAs)” —contains information about the RAS warranty and obtaining a return merchandise authorization (RMA).

Contact information

Patton Electronics offers a wide array of free technical services. If you have questions about any of our other products we recommend you begin your search for answers by using our technical knowledge base. Here, we have gathered together many of the more commonly asked questions and compiled them into a searchable database to help you quickly solve your problems.



Consumers that received the SmartLink in conjunction with telephony services from a VoIP service provider are encouraged to use the service provider as a primary point of contact for resolving problems.

Patton support headquarters in the USA

- Online support: available at **www.patton.com**
- E-mail support: e-mail sent to **support@patton.com** will be answered within 1 business day
- Telephone support: standard telephone support is available five days a week—from **8:00 am to 5:00 pm EST (1300 to 2200 UTC/GMT)**—by calling **+1 (301) 975-1007**
- Toll-Free VoIP support: call **sip:support@patton.com** with a VoIP SIP client
- Fax: **+1 (253) 663-5693**

Alternate Patton support for Europe, Middle East, and Africa (EMEA)

- Online support: available at **www.patton-inalp.com**
- E-mail support: e-mail sent to **support@patton-inalp.com** will be answered within 1 business day
- Telephone support: standard telephone support is available five days a week—from **8:00 am to 5:00 pm CET (0900 to 1800 UTC/GMT)**—by calling **+41 (0)31 985 25 55**
- Toll-Free VoIP support: call **sip:support@patton.com** with a VoIP SIP client
- Fax: **+41 (0)31 985 25 26**

Warranty Service and Returned Merchandise Authorizations (RMAs)

Patton Electronics is an ISO-9001 certified manufacturer and our products are carefully tested before shipment. All of our products are backed by a comprehensive warranty program.

Note If you purchased your equipment from a Patton Electronics reseller, ask your reseller how you should proceed with warranty service. It is often more con-

venient for you to work with your local reseller to obtain a replacement. Patton services our products no matter how you acquired them.

Warranty coverage

Our products are under warranty to be free from defects, and we will, at our option, repair or replace the product should it fail within one year from the first date of shipment. Our warranty is limited to defects in workmanship or materials, and does not cover customer damage, lightning or power surge damage, abuse, or unauthorized modification.

Out-of-warranty service

Patton services what we sell, no matter how you acquired it, including malfunctioning products that are no longer under warranty. Our products have a flat fee for repairs. Units damaged by lightning or other catastrophes may require replacement.

Returns for credit

Customer satisfaction is important to us, therefore any product may be returned with authorization within 30 days from the shipment date for a full credit of the purchase price. If you have ordered the wrong equipment or you are dissatisfied in any way, please contact us to request an RMA number to accept your return. Patton is not responsible for equipment returned without a Return Authorization.

Return for credit policy

- Less than 30 days: No Charge. Your credit will be issued upon receipt and inspection of the equipment.
- 30 to 60 days: We will add a 20% restocking charge (crediting your account with 80% of the purchase price).
- Over 60 days: Products will be accepted for repairs only.

RMA numbers

RMA numbers are required for all product returns. You can obtain an RMA by doing one of the following:

- Completing a request on the RMA Request page in the *Support* section at **www.patton.com**
- By calling **+1 (301) 975-1007** and speaking to a Technical Support Engineer
- By sending an e-mail to **returns@patton.com**

All returned units must have the RMA number clearly visible on the outside of the shipping container. Please use the original packing material that the device came in or pack the unit securely to avoid damage during shipping.

Shipping instructions

The RMA number should be clearly visible on the address label. Our shipping address is as follows:

Patton Electronics Company

RMA#: xxxx

7622 Rickenbacker Dr.

Gaithersburg, MD 20879-4773 USA

Patton will ship the equipment back to you in the same manner you ship it to us. Patton will pay the return shipping costs.

Appendix A **Compliance information**

Chapter contents

Compliance	76
EMC Compliance:	76
Safety Compliance	76
Radio and TV Interference (FCC Part 15)	76
CE Notice (Declarations of Conformity).....	76

Compliance

EMC Compliance:

FCC Part 15, Class B

EN55022, Class B

EN55024

Safety Compliance

EN60950-1

Radio and TV Interference (FCC Part 15)

This equipment generates and uses radio frequency energy, and if not installed and used properly—that is, in strict accordance with the manufacturer's instructions—may cause interference to radio and television reception. This equipment has been tested and found to comply with the limits for a Class B computing device in accordance with the specifications in Subpart B of Part 15 of FCC rules, which are designed to provide reasonable protection from such interference in a commercial installation. However, there is no guarantee that interference will not occur in a particular installation. If the equipment causes interference to radio or television reception, which can be determined by disconnecting the cables, try to correct the interference by one or more of the following measures: moving the computing equipment away from the receiver, re-orienting the receiving antenna, and/or plugging the receiving equipment into a different AC outlet (such that the computing equipment and receiver are on different branches).

CE Notice (Declarations of Conformity)

We certify that the apparatus identified in this document conforms to the requirements of Council Directive 1999/5/EC on the approximation of the laws of the member states relating to Radio and Telecommunication Terminal Equipment and the mutual recognition of their conformity.

The safety advice in the documentation accompanying this product shall be obeyed. The conformity to the above directive is indicated by the CE sign on the device.

Appendix B **Specifications**

Chapter contents

Voice Connectivity	78
Connectivity	78
Voice Processing (signalling dependent)	78
Fax and Modem Support	78
Voice Services/Features	79
IP Services	79
Management	79
Operating Environment	79
System	79

Voice Connectivity

2-wire Loopstart, RJ-11/12

Short haul loop 1.1 km @3REN

Caller-ID Type-1/2 FSK and ITU V.23/Bell 202 generation

Connectivity

1 10/100 Full Duplex/Autosensing Ethernet RJ-45

Voice Processing (signalling dependent)

SIP

MGCP

- Packet Cable NCS 1.0
- IETF MGCP 1.0

Voice codes

- G.711 A-Law/ μ -Law (64 kbps)
- G.726 (ADPCM 40, 32, 24, 16 kbps)
- G.723.1 (5.3 or 6.3 kbps)
- G.729ab (8 kbps)

G.168 echo cancellation

2 parallel voice connections

DTMF detection and generation

Carrier tone detection and generation

Silence suppression and comfort noise

Configurable dejitter buffer

DTFM in-band & out-of-band

Configurable transmit packet length

RTP/RTCP (RFC 1889)

Fax and Modem Support

G.711 transparent FAX

T.38 Fax relay (9.6 k, 14.4 k)

Voice Services/Features

Call forwarding
Call transfer
Call hold
Call waiting
3-way calling

IP Services

IPv4 M-ATA; RIPv1, v2 (RFC 1058 and 2453)
IP filtering
NAPT
NTP
DHCP client & server
PPPoE
ICMP redirect (RFC 792); Packet fragmentation
DiffServe/ToS set or queue per header bits

Management

Browser configuration interface
Voice prompt configuration
TFTP configuration & firmware loading
SNMP v2 agent (MIB II and private MIB)

Operating Environment

Operating temperature: 0–40°C (32–104°F)
Operating humidity: 5–80% (non condensing)

System

Power: 100–240 VAC (50/60 Hz)

Appendix C **Dial plans**

Chapter contents

Introduction	81
Sample Dial Plans.....	81
Simple Dial Plan	81
Non-dialed Line Dial Plan	81
Complex Dial Plan	81

Introduction

The MGCP and SIP code will allow provisioning (via web browser) of the dial plan. A dial plan gives the unit a map to determine when a complete number has been entered and should be passed to the SIP server or gatekeeper for resolution into a destination IP address. Dial plans are expressed using the same syntax as used by MGCP NCS specification.

The formal syntax of the dial plan is described by the following notation:

```
Digit ::= "0" | "1" | "2" | "3" | "4" | "5" | "6" | "7" | "8" | "9"
Timer ::= "T" | "t"
Letter ::= Digit | Timer | "#" | "*" | "A" | "a" | "B" | "b" | "C" | "c" | "D" | "d"
Range ::= "X" | "x" -- matches any digit
| "[" Letters "]" -- matches any of the specified letters
Letters ::= Subrange | Subrange Letters
Subrange ::= Letter -- matches the specified letter
| Digit "-" Digit -- matches any digit between first and last
Position ::= Letter | Range
StringElement ::= Position -- matches any occurrence of the position
| Position "." -- matches an arbitrary number of occurrences
including 0
String ::= StringElement | StringElement String
StringList ::= String | String "|" StringList
DialPlan ::= String | "(" StringList ")"
```

A dial plan, according to this syntax, is defined either by a (case insensitive) string or by a list of strings. Regardless of the above syntax a timer is only allowed if it appears in the last position in a string (12T3 is not valid). Each string is an alternate numbering scheme. The unit will process the dial plan by comparing the current dial string against the dial plan, if the result is under qualified (partial matches at least one entry) then it will do nothing further. If the result matches or is over-qualified (no further digits could possibly produce a match) then send the string to the gatekeeper and clear the dial string. The Timer T is activated when it is all that is required to produce a match. The period of timer T is 4 seconds. For example a dial plan of (xxxT|xxxxx) will match immediately if 5 digits are entered, it will also match after a 4 second pause when 3 digits are entered.

Sample Dial Plans

Simple Dial Plan

Allows dialing of 7-digit numbers (e.g. 5551234) or an operator on 0. Dial plan is (0T|xxxxxxx)

Non-dialed Line Dial Plan

As soon as handset is lifted the unit contacts the gatekeeper (used for systems where DTMF detection is done in-call). Dial plan is (x.) i.e. match against 0 (or more) digits. Note: the dot ‘.’

Complex Dial Plan

- Local operator on 0, long distance operator on 00
- 4-digit local extension number starting with 3, 4, or 5
- 7-digit local numbers are prefixed by an 8

- 2-digit star services (e.g. 69)
- 10-digit long distance prefixed by 91
- International numbers starting with 9011+variable number of digits.

Dial plan for this is:

```
(0T|00T|[3-5]xxx|8xxxxxxx|*xx|91xxxxxxxxxx|9011x.T)
```

Appendix D **Calling Features**

Chapter contents

Introduction.....	84
-------------------	----

Introduction

The SmartLink M-ATA family supports advanced calling features that can be turned on and off from phones attached to the SmartLink M-ATA (see [table 3](#)).

Note Your telephony service provider must enable your service for these calling features to work.

Note F in [table 3](#) refers to the *hook flash* event.

Table 3. Calling features

Feature	Keypad	Feature	Keypad
Call Hold	F1	Call Retrieve	F*
Conference	F7	Conference Drop	F8
Call Transfer	F4		
Do not Disturb ON	*82	Do not Disturb OFF	#82
Distinctive ON	*90	Distinctive OFF	#90
Call Waiting ON	*91	Call Waiting OFF	#91
Incoming Caller ID Display ON	*92	Incoming Caller ID Display OFF	#92
Self Caller ID Block Service ON	*93	Self Caller ID Block Service OFF	#93
Anonymous Call Reject ON	*94	Anonymous Call Reject OFF	#94
Incoming Call Block ON	*95	Incoming Call Block OFF	#95
Call Forward Selective ON	*96	Call Forward Selective OFF	#96
Call Forward All ON	*97	Call Forward All OFF	#97
Call Forward Busy ON	*98	Call Forward Busy OFF	#98
Warm Line ON	*99	Warm Line OFF	#99
IP Dialing ON	*80	IP Dialing OFF	#80
Speed Dialing ON	*81	Speed Dialing OFF	#81
Call Return	*60		
Config Warm Line Number (*70yyyyy where yyyyy = number to call)	*70		
Config Speed Dialing Number (*71xyyyyy where x = speed dial key and yyyyy = number to call)	*71		
Config IP Dialing (*72xxx*xxx*xxx*xxx*yyyy where xxx = IP address and yyyy = optional port number)	*72		
Set Call Forward Number (Wait for 3 short confirmation tones before hanging up)	*73		
Access Voicemail	*86		

Appendix E **Voice prompt configuration**

Chapter contents

Introduction	86
Accessing the voice prompt	86
Existing voice prompt configuration	86

Introduction

The M-ATA provides the ability to review and set the network configuration parameters using the handset of an attached analog telephone handset.

By default from the factory, DHCP is enabled and an IP address is not configured.

The M-ATA must be power cycled or reloaded after changing any of the network settings. Menu selection item *Network Status* will not reflect setting changes until after M-ATA is reloaded or power cycled.

Note Configuration of these settings can result in loss of connectivity to the M-ATA on the local LAN.

Accessing the voice prompt

Dial **** from the analog handset to reach the main menu.

Existing voice prompt configuration

On hook the analog phone.

Access Code	Main Menu Selection	Announcement/ Function	Voice Prompt	User input
****	Main Menu	Plays main menu selections	Patton SmartLink Configuration Main Menu	Enter selection code
100#	Network status	Plays DHCP setting, IP address, gateway IP address and IP network mask setting	100# Network status	None.
110#	DHCP setting	Enables or Disables DHCP	110# DHCP Settings	1# to enable DHCP 2# to disable DHCP or "#" to return to the main menu
120#	IP address setting	Sets IP address of the M-ATA	120# Set IP Address	Use "*" to instead of ".", and "#" to end. Ex: 172*16*230*227# or "#" to return to the main menu
130#	Gateway setting	Sets the gateway router IP address	130# Set gateway router IP address	Use "*" to instant of ".", and "#" to end or "#" to return to the main menu
140#	Net mask setting	Set the IP network mask	140# Set IP network mask	Use "*" to instant of ".", and "#" to end or "#" to return to the main menu
150#	Reload	Immediately reloads the SmartLink	None	None

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