

Router

Voice

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Product Information

Product Name:	SPA9000	Serial Number:	FM700J902734
Software Version:	6.1.5	Hardware Version:	2.0.5
MAC Address:	000E08E4E353	Client Certificate:	Installed
Customization:	Open	Licenses:	None

System Status

Current Time:	2/24/2015 09:01:00	Elapsed Time:	11:41:12
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FXS 1 Status

Hook State:	On	Message Waiting:	No
Call Back Active:	No	Last Called Number:	
Last Caller Number:			
Call 1 State:	Idle	Call 2 State:	Idle
Call 1 Tone:	None	Call 2 Tone:	None
Call 1 Encoder:		Call 2 Encoder:	
Call 1 Decoder:		Call 2 Decoder:	
Call 1 FAX:		Call 2 FAX:	
Call 1 Type:		Call 2 Type:	
Call 1 Remote Hold:		Call 2 Remote Hold:	
Call 1 Callback:		Call 2 Callback:	
Call 1 Peer Name:		Call 2 Peer Name:	
Call 1 Peer Phone:		Call 2 Peer Phone:	
Call 1 Duration:		Call 2 Duration:	
Call 1 Packets Sent:		Call 2 Packets Sent:	
Call 1 Packets Recv:		Call 2 Packets Recv:	
Call 1 Bytes Sent:		Call 2 Bytes Sent:	
Call 1 Bytes Recv:		Call 2 Bytes Recv:	

Line 1 Status

Registration State:	Registered	Last Registration At:	2/24/2015 08:50:24
Next Registration In:	257 s	Message Waiting:	No
Mapped SIP Port:			

Line 2 Status

Registration State:	Not Registered	Last Registration At:	
Next Registration In:		Message Waiting:	No
Mapped SIP Port:			

Line 3 Status

Registration State:	Registered	Last Registration At:	2/24/2015 08:50:24
Next Registration In:	257 s	Message Waiting:	No
Mapped SIP Port:			

Line 4 Status

Registration State:	Registered	Last Registration At:	2/24/2015 08:21:50
Next Registration In:	1219 s	Message Waiting:	No
Mapped SIP Port:			

Auto Attendant Prompt Status

Prompt 1:	28875 ms	Prompt 2:	2625 ms
Prompt 3:	5250 ms	Prompt 4:	15750 ms
Prompt 5:	5250 ms	Prompt 6:	10500 ms
Prompt 7:	5250 ms	Prompt 8:	13125 ms
Prompt 9:	0 ms	Prompt 10:	5250 ms
Space Remaining:	2625 ms	Current AA:	Daytime

Internal Music Status

Installed Music Path:	Factory Default
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SIP Parameters

Max Forward:	70	Max Redirection:	5
Max Auth:	2	SIP User Agent Name:	\$VERSION
SIP Server Name:	\$VERSION	SIP Reg User Agent Name:	
SIP Accept Language:		DTMF Relay MIME Type:	application/dtmf-relay
Hook Flash MIME Type:	application/hook-flash	Remove Last Reg:	no
Use Compact Header:	no	Escape Display Name:	yes
RFC 2543 Call Hold:	yes	Mark All AVT Packets:	yes
SIP TCP Port Min:	5060	SIP TCP Port Max:	5060

SIP Timer Values (sec)

SIP T1:	.5	SIP T2:	4
SIP T4:	5	SIP Timer B:	32
SIP Timer F:	32	SIP Timer H:	32
SIP Timer D:	32	SIP Timer J:	32
INVITE Expires:	240	ReINVITE Expires:	30
Reg Min Expires:	1	Reg Max Expires:	7200
Reg Retry Intvl:	30	Reg Retry Long Intvl:	1200
Reg Retry Random Delay:		Reg Retry Long Random Delay:	
Reg Retry Intvl Cap:			

Response Status Code Handling

SIT1 RSC:		SIT2 RSC:	
SIT3 RSC:		SIT4 RSC:	
Try Backup RSC:		Retry Reg RSC:	

RTP Parameters

RTP Port Min:	16384	RTP Port Max:	16482
RTP Packet Size:	0.020	Max RTP ICMP Err:	0

RTCP Tx Interval:	0	No UDP Checksum:	no
Stats In BYE:	no		

SDP Payload Types

NSE Dynamic Payload:	100	AVT Dynamic Payload:	101
INFOREQ Dynamic Payload:		G726r16 Dynamic Payload:	98
G726r24 Dynamic Payload:	97	G726r32 Dynamic Payload:	2
G726r40 Dynamic Payload:	96	G729b Dynamic Payload:	99
EncapRTP Dynamic Payload:	112	RTP-Start-Loopback Dynamic Payload:	113
RTP-Start-Loopback Codec:	G711u	NSE Codec Name:	NSE
AVT Codec Name:	telephone-event	G711u Codec Name:	PCMU
G711a Codec Name:	PCMA	G726r16 Codec Name:	G726-16
G726r24 Codec Name:	G726-24	G726r32 Codec Name:	G726-32
G726r40 Codec Name:	G726-40	G729a Codec Name:	G729a
G729b Codec Name:	G729ab	G723 Codec Name:	G723
EncapRTP Codec Name:	encaprtp		

NAT Support Parameters

Handle VIA received:	no	Handle VIA rport:	no
Insert VIA received:	no	Insert VIA rport:	no
Substitute VIA Addr:	no	Send Resp To Src Port:	no
STUN Enable:	no	STUN Test Enable:	no
STUN Server:		EXT IP:	
EXT RTP Port Min:		NAT Keep Alive Intvl:	15

PBX Parameters

Proxy Network Interface:	WAN	Proxy Listen Port:	6060
Multicast Address:	224.168.168.168:6061	Group Page Address:	224.168.168.168:34567
Max Expires:	60	Force Media Proxy:	yes
Proxy Debug Option:	full		

Call Routing Rule:	<input type="text" value="(<:L4>9xx. <:L2>8xx.)"/>		
Call Park MOH Server:	<input type="text" value="imusic"/>	Call Park DLG Refresh Intvl:	<input type="text" value="0"/>
Default Group Line:	<input type="text" value="1,2,3,4"/>	Group 1 User ID:	<input type="text"/>
Group 1 Line:	<input type="text"/>	Group 2 User ID:	<input type="text"/>
Group 2 Line:	<input type="text"/>	Group 3 User ID:	<input type="text"/>
Group 3 Line:	<input type="text"/>	Group 4 User ID:	<input type="text"/>
Group 4 Line:	<input type="text"/>		
Hunt Groups:	<input type="text" value="500:name='Obchod',102,105,hunt=al;14;1,cfwd=105 501:name='Reklamac"/>		
SIP DIDN Field:	<input type="text" value="TO UserID"/>	SIP DIDN Param Name:	<input type="text" value="didn"/>
Accept All MWI As Line:	<input type="text" value="Current"/>	Phone DLG Refresh Intvl:	<input type="text" value="0"/>

Internal Music Source Parameters

Internal Music URL:	<input type="text"/>		
Internal Music Script:	<input type="text"/>		
Internal MOH Refresh Intvl:	<input type="text" value="0"/>	Internal Music LBR Codec:	<input type="text" value="G729a"/>
Internal Music Preferred Codec:	<input type="text" value="G711u"/>	Internal Music Use Pref Codec Only:	<input type="text" value="no"/>

Auto Attendant Parameters

AA Dial Plan 1:	<input type="text" value="(<0:101> <1:105> <2:108> <3:108> <4:108> <5:101> <6:105> <8:108"/>		
AA Dial Plan 2:	<input type="text" value="(<0:101> <1:105> <2:108> <3:108> <4:108> <5:101> <6:105> <8:108"/>		
AA script 1:	<input type="text" value="<aa><form id='dir' type='menu'> <audio src='prompt1' bargein='T'/> <n"/>		
AA script 2:	<input type="text" value="<aa><form id='dir' type='menu'> <audio src='prompt4' bargein='T'/><au"/>		
AA script 3:	<input type="text"/>		
DayTime AA:	<input type="text" value="yes"/>	Day Time:	<input type="text" value="start=7:0:0;end=23:0:0"/>
DayTime AA Script:	<input type="text" value="1"/>	DayTime Answer Delay:	<input type="text" value="2"/>
NightTime AA:	<input type="text" value="yes"/>	NightTime AA Script:	<input type="text" value="2"/>
NightTime Answer Delay:	<input type="text" value="2"/>	Weekend/Holiday AA:	<input type="text" value="yes"/>
Weekends/Holidays:	<input type="text" value="wk=6,7"/>	Weekend/Holiday AA Script:	<input type="text" value="2"/>
Weekend/Holiday Answer Delay:	<input type="text" value="2"/>	AA LBR Codec:	<input type="text" value="G729a"/>
AA Preferred Codec:	<input type="text" value="G711u"/>	AA Use Pref Codec Only:	<input type="text" value="no"/>
AA Decode Inband DTMF:	<input type="text" value="yes"/>		
AA Prompts URL Script:	<input type="text" value="serv=http://192.168.2.102:29863/prompts/;p1=Prompt1.wav;p2=Prompt2.w"/>		

PBX Phone Parameters

Next Auto User ID:	<input type="text" value="100"/>	Phone Ext Password:	<input type="text"/>
Phone Upgrade Rule:	<input type="text"/>		
Phone Dial Plan:	<input type="text" value="(<*97:vmm> 9,[3469]11S0 9,[2-9]xxxxxx 9,[2-9]xxxxxxxxS0 9,1[2-9]xxx"/>		
Phone Config XML:	<input type="text"/>		
Use LVS_PROXY:	<input type="text" value="no"/>	CTI Enable:	<input type="text" value="no"/>

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