



Linksys Phone Adapter Configuration

Router

Voice

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Product Information

Product Name:	SPA-3102	Serial Number:	FM600F321552
Software Version:	5.2.13(GW002)	Hardware Version:	1.4.5
MAC Address:	000E08CC88DE	Client Certificate:	Installed
Customization:	Open		

System Status

Current Time:	3/12/2012 19:14:00	Elapsed Time:	00:10:15
RTP Packets Sent:	0	RTP Bytes Sent:	0
RTP Packets Recv:	0	RTP Bytes Recv:	0
SIP Messages Sent:	281	SIP Bytes Sent:	135634
SIP Messages Recv:	0	SIP Bytes Recv:	0
External IP:	86.96.105.77		

Line 1 Status

Hook State:	On	Registration State:	Failed
Last Registration At:	0/0/0 00:00:00	Next Registration In:	27 s
Message Waiting:	No	Call Back Active:	No
Last Called Number:		Last Caller Number:	
Mapped SIP Port:	5060		
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:	
Call 1 Decode Latency:		Call 2 Decode Latency:	
Call 1 Jitter:		Call 2 Jitter:	
Call 1 Round Trip Delay:		Call 2 Round Trip Delay:	
Call 1 Packets Lost:		Call 2 Packets Lost:	
Call 1 Packet Error:		Call 2 Packet Error:	
Call 1 Mapped RTP Port:		Call 2 Mapped RTP Port:	
Call 1 Media Loopback:		Call 2 Media Loopback:	

PSTN Line Status

Hook State:	On	Line Voltage:	0 (V)
Loop Current:	0.0 (mA)	Registration State:	Failed
Last Registration At:	0/0/0 00:00:00	Next Registration In:	28 s
Last Called VoIP Number:		Last Called PSTN Number:	

Last VoIP Caller:		Last PSTN Caller:	
Last PSTN Disconnect Reason:		PSTN Activity Timer:	30000 (ms)
Mapped SIP Port:	5061	Call Type:	
PSTN State:	Idle	VoIP State:	Idle
PSTN Tone:		VoIP Tone:	
PSTN Peer Name:		VoIP Peer Name:	
PSTN Peer Number:		VoIP Peer Number:	
VoIP Call Decoder:		VoIP Call Encoder:	
VoIP Call Remote Hold:		VoIP Call FAX:	
VoIP Call Packets Sent:		VoIP Call Duration:	
VoIP Call Bytes Sent:		VoIP Call Packets Recv:	
VoIP Call Decode Latency:		VoIP Call Bytes Recv:	
VoIP Call Round Trip Delay:		VoIP Call Jitter:	
VoIP Call Packet Error:		VoIP Call Packets Lost:	
		VoIP Call Mapped RTP Port:	

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