



IP Phone SPA525G2

Cisco Systems, Inc.

Voice Wi-Fi Bluetooth Personal Address Book Call History Speed Dials Firmware Upgrade

Info System **SIP** Provisioning Regional Phone Ext 1 Ext 2 Ext 3 Ext 4 Ext 5 User Att Console TR-069

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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 <input type="checkbox"/>
Use Compact Header:	no <input type="checkbox"/>	Escape Display Name:	no <input type="checkbox"/>
SIP-B Enable:	no <input type="checkbox"/>	Talk Package:	no <input type="checkbox"/>
Hold Package:	no <input type="checkbox"/>	Conference Package:	no <input type="checkbox"/>
Notify Conference:	no <input type="checkbox"/>	RFC 2543 Call Hold:	yes <input type="checkbox"/>
Random REG CID On Reboot:	no <input type="checkbox"/>	Mark All AVT Packets:	yes <input type="checkbox"/>
SIP TCP Port Min:	5060	SIP TCP Port Max:	5080
CTI Enable:	no <input type="checkbox"/>	Caller ID Header:	PAID-RPID-FROM <input type="checkbox"/>
SRTP Method:	x-sipura <input type="checkbox"/>	Hold Target Before REFER:	no <input type="checkbox"/>
Dialog SDP Enable:	no <input type="checkbox"/>	Keep Referee When REFER Failed:	yes <input type="checkbox"/>
Display Diversion Info:	no <input type="checkbox"/>	Display Anonymous From Header:	no <input type="checkbox"/>

SIP Timer Values (sec)

SIP T1:	.5	SIP T2:	4
SIP T4:	5	SIP Timer B:	16
SIP Timer F:	16	SIP Timer H:	16
SIP Timer D:	16	SIP Timer J:	16
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:		Sub Min Expires:	10
Sub Max Expires:	7200	Sub Retry Intvl:	10

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:	0.030	Max RTP ICMP Err:	0
RTCP Tx Interval:	0	No UDP Checksum:	no <input type="checkbox"/>
Symmetric RTP:	no <input type="checkbox"/>	Stats In BYE:	no <input type="checkbox"/>

SDP Payload Types

AVT Dynamic Payload:	101	INFOREQ Dynamic Payload:	
G726r32 Dynamic Payload:	2	G729b Dynamic Payload:	99
EncapRTP Dynamic Payload:	112	RTP-Start-Loopback Dynamic Payload:	113
RTP-Start-Loopback Codec:	G711u <input type="checkbox"/>	AVT Codec Name:	telephone-event
G711u Codec Name:	PCMU	G711a Codec Name:	PCMA
G726r32 Codec Name:	G726-32	G729a Codec Name:	G729a

G729b Codec Name: G729ab
 EncapRTP Codec Name: encaprtsp

G722 Codec Name: G722

NAT Support Parameters

Handle VIA received: no
 Insert VIA received: no
 Substitute VIA Addr: no
 STUN Enable: no
 STUN Server:
 EXT RTP Port Min:
 Redirect Keep Alive: no

Handle VIA rport: no
 Insert VIA rport: no
 Send Resp To Src Port: no
 STUN Test Enable: no
 EXT IP:
 NAT Keep Alive Intvl: 15

Linksys Key System Parameters

Linksys Key System: no
 Key System Auto Discovery: yes
 Force LAN Codec: none

Multicast Address: 224.168.168.168:6061
 Key System IP Address:
 Auto Ans GrPage On Active Call: yes

Undo All Changes

Submit All Changes

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