

Q-SYS Softphone SIP Integration Worksheet

Please fill out this form with the requested information. This will allow us to preconfigure the Q-SYS core and greatly simplify the process of integrating a Q-SYS softphone.

IP Address

Will this SIP endpoint be given a static IP address or will it be DHCP?

If static, please provide:

IP Address: _____

Net mask: _____

Either DHCP or static:

DNS Server, if needed: _____

Proxy information

Address or hostname of SIP Proxy: _____ (if host, specify DNS above)

SIP Transport (select one): TCP UDP TLS

SIP port to be used (5060 default): _____

SIP Domain: _____

Backup Proxy (optional): _____ (if host, specify DNS above)

Subscriber information

Subscriber Number: _____

Digest Username: _____

Digest Password: _____

Required Voice Codec Support (check all that apply)

- | | |
|-------------------------------------|--|
| <input type="checkbox"/> G.722 | <input type="checkbox"/> G.726 32k(AAL2) |
| <input type="checkbox"/> G.711 ulaw | <input type="checkbox"/> G.726 32K |
| <input type="checkbox"/> G.711 alaw | |

DTMF support (check all that apply)

- RFC2833 (RTP Event)
 SIP Info
 In-band DTMF only

Other settings (check all that apply)

- Enable Stun
 Enable SRTP

Notes:

- Certificate management not supported.
- 802.1Q VLAN tagging not supported.
- SIP QoS DSCP settings not supported, please use switch for marking DSCP settings based on UDP port numbers for SIP RTP media traffic if needed.