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)

- ☐ G.722
- ☐ G.711 ulaw
- ☐ G.711 alaw
- ☐ G.726 32k(AAL2)
- ☐ G.726 32k

**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.