

Preferences

TCP, TLS and Secure RTP Options

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We support both TCP (Transmission Control Protocol) and UDP (User Datagram Protocol) communication protocols. While UDP is by far the most common of the two protocols TCP is stated to have advantages.

TCP advantages

- **Keep Alives:** SIP must periodically send out keep-alives to maintain the NAT table entry. The required frequency of keep-alives is much higher for UDP (maybe every 30 seconds) vs TCP (maybe every 15 minutes). While not relevant for small installations TCP has significant advantages within enterprise installations.
- **TLS:** In a security conscious world TCP enables your end points with Transport Layer Security (TLS) which over the separate port 5061 instead of the normal 5060 for UDP.
- **SRTP (Secure Real-Time Transport Protocol or Secure RTP):** SRTP encrypts or “codes” the voice data itself so that no one can understand what is being said except the person who has the decoding “key”, or the person to whom the call is being made. SRTP and SIP TLS encrypt different parts of the VoIP service but together security conscious organisations such as financial or military to use this service.

If you choose to use Secure Encrypted RTP this may cause calls to fail if you have not configured SRTP on your client.

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Quick Guide

Step One: TCP, TLS & Secure RTP Options

1. Log into <https://portal.hero.co.nz> > Voice tab and select number you wish to set up remote call back on.
2. Select **Other Settings > TLS & Secure RTP Options.**
3. Select one of the SIP Transport options.
4. Enable Secure Encrypted RTP

SIP over TCP and TLS support for secure encrypted signalling on calls. SRTP is now also supported for encrypted calls.

SIP Transport Support for UDP, TCP and TLS (Default)
 Force Support for UDP Only
 Force Support for TCP Only
 Force Support for TLS Only

Force calls to use Secure Encrypted RTP? (WARNING: This may cause calls to fail if you have not configured SRTP on your client)

5. Click **Save settings** to update

Force calls to use Secure Encrypted RTP? (WARNING: This may cause calls to fail if you have not configured SRTP on your client)

Step Two: Change PBX/handset transport

1. Log into IP PBX / Handset
2. Select Account
3. Find option to change Transport > TLS

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