

CallPlus has been developing Voice over IP products since 2001, our latest product SIP trunking is designed to be a true PSTN or ISDN replacement product. Existing VoIP products in the market place use the internet to make and receive calls, the internet is a 'best efforts' transportation mechanism. CallPlus's SIP trunking service is either delivered over SHDSL.G or Metro Ethernet Fibre direct to our Core Network. This enables us to provide a 1:1 private dedicated link to our core network, enabling Quality of Service.

CallPlus SIP Trunking has four components:

1. Access (either Ethernet Fibre or SHDSL.G)
2. Transmission of voice traffic to our core network (1:1 contention ratio)
2. SIP trunks (\$15 per trunk / line / channel)
3. Additional services (DDI Blocks, Caller ID and DDI billing)

Devices that can be connected to our SIP Trunking Service:

- Compatible IP PABX (see our website for a full list)
- Asterisx box or Trixbox
- CISCO UC520 or Call Manager
- SIP Gateway device

We offer the following voice codecs:

- G711 - 1Mbps connection = 10 Channels / trunks
- G729 - 1Mbps connection = 25 Channels / trunks

CallPlus recommends G711

Porting existing numbers to SIP Trunking:

- You can port your existing numbers, CustomerLinks and DDI's.
- Porting charge = \$18 per DDI / number

Key features:

- Low line rental rates
- 1000 free local minutes per month (per account)
- Low calling rates
- Free internal extension dialing
- Purchase lines in single increments
- Unlimited interbranch Voice over IP calls
- Free online billing and reporting
- Convenience of one phone and internet bill

Please talk to your CallPlus Corporate Sales Consultant, Account Manager or call 0800 89 5000 to find out more.

