SIP Trunking is an innovative IP voice solution that provides businesses like yours with customized next-generation voice services. By replacing your costly PRIs and traditional analog lines, SIP Trunking simplifies your company’s telephony infrastructure and saves you money.

**WHY SIP TRUNKING IS CREATING A BUZZ**

- Complete your unified communications strategy and dramatically reduce expenses
  - Manage the same call capacity with fewer lines (or “channels”)
  - Enjoy Capacity On Demand - accommodate any call volume by only paying for what you need, when you need it
  - Create a local presence and eliminate long distance charges by selecting local phone numbers in over 1000 cities across Canada
- Make changes to your account instantly thanks to sophisticated online management tools
- Increase efficiency by pooling and seamlessly shifting capacity across multiple sites
- Bank on our redundant national voice network build on carrier-class technology

**IS SIP TRUNKING RIGHT FOR YOUR BUSINESS?**

- Do you have an IP-ready PBX?
- Are you thinking about purchasing a new PBX?
- Do you have multiple office sites?
- Do you experience highly variable call volume?
- Do you ever experience issues with busy signals or insufficient lines?
- Are you seeking enterprise-level reliability and scalability?
Overview of How SIP Trunking Configuration Differs from a Traditional Fixed-Line Setup

**NETWORK SPECIFICATIONS AND CAPABILITIES**

- Up to eight IP addresses per SIP Trunk for built-in redundancy
- Codec support for G.711, G.726, G.721, and G.729a/b
- QoS support via TOS bit
- Support for IPSec VPN
- T38 fax support
- Fully redundant class 4/5 core switching
- Directory listings for all DIDs
- 10 or 11 digit dialling (e.164)
- 15 character Caller ID and Caller Name
- Basic and VoIP V911 with user-managed location for each DID
- Private IP Cross Connection availability
- Certified Cisco CallManager and UC 520/540/560PBXs
- Certified directly with Microsoft OCS R2 and Lync

**WHAT YOU NEED TO SUCCESSFULLY DEPLOY A SIP TRUNK:**

- Sufficient bandwidth to handle your voice needs
- An IP-PBX with a SIP-enabled trunk side
- A SIP-ready enterprise edge device

FOR ADDITIONAL INFORMATION PLEASE CONTACT US AT 644-4482.