

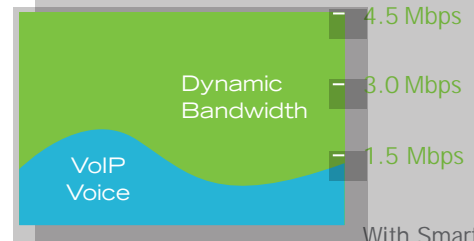
SmartVoice SIP Trunking

Do your voice and data needs continuously fluctuate? Now you can take advantage of employee calling patterns to raise your data speeds. Using the dynamic capabilities of SIP technology and “bonding,” your business can get the voice services it needs while increasing the data bandwidth available to improve employee productivity.

SmartVoice SIP Trunking offers a SIP access trunk to your premises with a PRI or CAS network connection for your PBX, enabling you to use your existing digital equipment or the latest in IP PBXs.

Up to 4.5 Mbps throughput

Using Multi-link Point to Point Protocol, we are able to “bond” T1s together to provide a throughput of up to 4.5 Mbps of bandwidth in NxT1 increments. When employees are not making phone calls, the bonded access facility is fully available for data traffic. Of course, voice takes priority so when an employee makes or receives a call, the SmartVoice trunk immediately allocates bandwidth to the voice traffic, but only as long as the call is in process. Once the call is over, the bandwidth is again available for data traffic. This means, as your voice needs fluctuate throughout the day, your data bandwidth adjusts to capitalize on the available capacity.

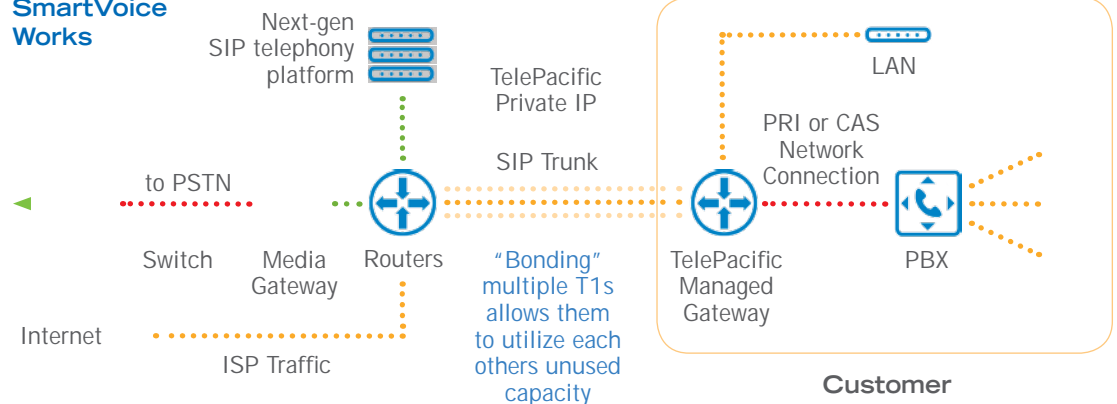


With SmartVoice, when voice lines aren't in use, bandwidth can be shifted to data traffic.



SmartVoice SIP Trunks offer a PRI or a CAS network connection with dynamic voice and data integration in NxT1 increments up to 4.5Mbps, tripling the effective access bandwidth with optimum throughput and efficiency.

How SmartVoice Works



SmartVoice SIP Trunks use an advanced telephony platform for call control and voice features. This redundant platform resides in the core of our private IP network, delivering SIP services from your premises to the PSTN. Likewise, the SIP based trunk terminates at your premises and is converted to traditional call control protocols — PRI or CAS — with a TelePacific managed gateway.

With SmartVoice, you get...

- ▶ Advanced SIP trunks with a PRI or CAS network connection provided by a TelePacific managed premise gateway
- ▶ Dynamically integrated, high capacity IP access on our private IP network, which prioritizes voice from your premises to the PSTN
- ▶ Data services over a fully meshed IP network with Service Level Agreements offering up to 99.999% uptime.
- ▶ Flexible outbound calling packages with generous usage amounts
- ▶ The key voice and data features you expect

Choose Your Voice Services

SmartVoice SIP Trunks with a CAS or PRI network connection offer the power and performance of high-speed IP networking to meet today's business needs. This high capacity, end-to-end IP connection allows between 8 to 36 simultaneous voice calls with CAS, and between 8 and 34 simultaneous calls with PRI.

Confidence in your data service

Your Internet connection is only a click away with our Dedicated Internet Access (DIA). With either a single T1 —or two or three bonded T1s— as part of your service, your business has up to 4.5M at its fingertips.

Included in Your Package:

OneCentral	An online service portal that allows you to examine your bill, generate custom views, and access tools for performing in-depth statement analysis with downloadable content.
Fax Central	Three fax-to-email boxes. Access faxes anywhere you access email. There are no busy signals when people try to fax you, and confidential faxes go straight to your email box.
IP Addresses	8 IP addresses standard. But we offer up to 256 IP addresses based on utilization at no additional charge.
Email Hosting	Email hosting for up to 100 email boxes. More boxes are available a la carte.
Web Hosting	200 Mbps of disk space is provided for website hosting
Domain Name	Domain DNS and reverse DNS services including domain name transfer, hosting, and registration for one domain name.
Caller ID	Caller ID provides visibility to inbound caller information necessary for call screening and screen pop-up applications.
DIDs	100 Direct Inward Dial (DID) numbers. Give each employee their own number so that people can dial each other directly. Avoid the need for inbound calls to go through a receptionist.



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