



Harness the power of flexible connections

SIP Trunking

A Better Connection for Your Business Communications

SIP Trunking replaces the physical connections to your communications system with less expensive and more flexible SIP connections. SIP Trunking eliminates the need for traditional analog, T1 or PRI lines and moves your business into the advanced world of IP communications.

Savings & Scalability Today Enhanced Services Tomorrow

LOWER COSTS | GREATER FLEXIBILITY | FUTURE-PROOF

- + Right-size Your Costs**
SIP Trunking is a future-proof solution that expands in lockstep with your business. Unlike traditional PRI connections, our service lets you buy exactly the number of call paths you need, so you can right-size your costs.
- + Quickly Scale Your Call Capacity**
With traditional PRIs, adding capacity takes time because physical connections must be installed. Our SIP Trunking service can be increased to any capacity that the IP connection will support within hours or less if needed.
- + Leverage Your Existing PBX**
Our SIP Trunking service has been tested for interoperability with all major SIP-enabled PBX brands. Not on our list or need to keep a traditional connection to your PBX? Gateways can connect your PBX to our service.
- + Ensure Business Continuity**
In the event of an emergency or when all lines are busy, SIP Trunking allows you to immediately and easily reroute traffic to another location or phone number, so you never miss a call.
- + Easily Upgrade to Cloud Solutions**
SIP Trunking preserves your PBX investment while easing a future upgrade to a hosted PBX service. With SIP Trunking, you can take advantage of cloud services such as call recording immediately.
- + Get Unlimited Calling with Predictable Costs**
With SIP Trunking, you get predictable monthly costs and unlimited local and long distance calling. Plus, we offer competitive international rates when needed.



SIP Trunking

Less Cost, More Flexibility

Often the tradeoff for lower-cost services is reduced functionality. Not with our SIP Trunking service – you get the quality of a traditional connection plus a unique set of features. That’s because SIP (Session Initiation Protocol) supports voice calls, conference calls, multimedia and more. Plus, SIP transmits call-related info, such as caller ID, enabling enhanced functionality.



Keep Your Phone Numbers

Keep the phone numbers your customers know even if you move to a new area code.



Route Calls Between Locations

We offer flexible geographical routing and call balancing between your locations.



Understand Your Usage

Monitor your capacity patterns and usage so you can adjust capacities to your changing needs.



Record Your Calls

With our optional advanced recording capability, you can record and store calls, and easily access them when needed.



Stay Connected to E911

Get complete support for both 911 calling and Enhanced 911 (E911), so you can access emergency services.



Choose Your Calling ID

Determine how customers see your phone number – a single companywide number or each individual phone number.



Ensure Business Continuity

Calls to your main number – and every DID – can be forwarded to another location, mobile numbers, an answering service or voice mail in the event of power loss or Internet failure.



Establish Virtual Local Presence

Obtain local phone numbers for any areas in which you want to project local presence even if they’re out of your area. Calls to these numbers won’t incur long distance charges for you or your incoming caller.



Forward Calls When Busy

Send calls to alternate numbers when your PBX can’t be reached because all SIP Trunks are busy.



Experience a High-Quality Network

Rest assured that our service is delivered over a carrier-class network based on technology from Metaswitch.



Multisite SIP Trunk Rollover

Calls to SIP Trunks can overflow from one location to another, maximizing capacity across your organization.



Do Business As Usual

Reach customers and suppliers as usual through our network of more than 12,000 local calling areas in the US and Canada.

NEED TO MIGRATE TO THE CLOUD BUT NOT READY TO MOVE EVERYTHING? USE OUR CAP & GROW SOLUTION

Pursue a cap-and-grow strategy by networking your SIP-enabled premise PBX with our hosted PBX solution. Add new users to the hosted solution to stop capex investment in hardware that’s nearing end of life.

Deploy our Hosted/SIP Networking solution at a single site or multiple branches and you can still dial by extension and transfer calls intercompany while migrating to the cloud.

FOR MORE INFORMATION

 www.timetechnology.com  877.529.5220