



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.

+ 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.

+ Easily Upgrade to Cloud Solutions

SIP Trunking preserves your PBX investment while easing a future upgrade to a hosted PBX service. Or, cap hardware spend and add cloud with our hybrid SIP-Hosted PBX solution.

+ 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.

+ 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.

+ 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.

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 (and are in your advertising) even if you move to a new area code.



Understand Your Usage

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



Stay Connected to E911

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



Forward Calls When Busy

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



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.



Enable a Hybrid SIP-Hosted PBX

Begin your move to cloud by networking your SIP-enabled on-premises PBX with our hosted PBX at branch offices, enabling dial by extension and call transfer companywide.



Route & Rollover Calls by Location

Route your calls by geography and/or send overflow calls to another location, maximizing capacity across your organization.



Choose Your Calling ID

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



Experience a High-Quality Network

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



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.



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.



Record Your Calls

Enable advanced call recording by individual DID with this optional feature. Search and playback recordings through a simple web-based interface.

SIP TRUNKING COMMPORTAL ADMIN

Get control of critical capabilities with our web-based SIP Trunking CommPortal Admin. Uniquely, all of these services are enabled at the network level, ahead of your PBX, for improved reliability, business continuity and assured compliance with the latest 911 rules.

- ✓ Call Analytics
- ✓ Call Forwarding
- ✓ Account Code
- ✓ Emergency Call Notification
- ✓ Selective Call Rejection
- ✓ Short Code Configuration

FOR MORE INFORMATION

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