



SIP TRUNKING

A NEW WORLD OF
COMMUNICATION POSSIBILITIES

UPGRADE YOUR EXISTING PHONE SYSTEM WITH SIP TRUNKING.

Are you ready to upgrade your office communications and still use your current phone system? Like many small and medium businesses, you may be using two separate connections to connect your business to the world, one for voice and one for internet access. SIP Trunking is a way to efficiently use one connection for all your voice and data needs. This single connection will provide you with:

- High-quality, reliable phone service
- Internet access
- Dynamic allocation of capacity between voice and data services
- A suite of optional advanced features that are easily managed online

Now you can save money and boost productivity with useful new calling features too.

GET THE BENEFITS OF VOIP

SIP Trunking uses Voice Over Internet Protocol (VOIP) and our state-of-the-art network.

- SIP Trunking is available as digital trunks or as ISDN-PRI lines.
- It is compatible with your existing analog phones, key systems, PBX and with select IP PBX equipment.
- It will help you save money compared to your current local, long distance and internet services.

VOICE AND DATA SHARE A CIRCUIT

With SIP Trunking, voice and data are carried over a single circuit, with part used for data and part for voice. When your voice traffic is low, the circuit gives most of the bandwidth to data. As soon as a call is placed, voice traffic gets priority and the circuit will dynamically allocate more bandwidth to voice to ensure call quality. It's efficient and simple to manage.



EASY TO ADD EXTRA CALLING FEATURES

All your existing PBX features will work as they did before. And with SIP Trunking you can start to take advantage of useful extra features such as:

- Individual user control of incoming call routing
- Automated attendant
- Simultaneous ringing of desk phones and mobiles
- Back-up services during power outages

END-TO-END SECURE & RELIABLE SOLUTION

- We deliver this high-quality, reliable SIP Trunking solution from our managed network to your premises.
- We'll help you plan a seamless transition to VOIP.
- Our skilled local support team is always available along with our round-the-clock network operations team.

With SIP trunking, you can enter a new world of communication possibilities.

TAKE CONTROL WITH COMMPORTAL

CommPortal offers a full suite of tools designed to help you better manage your communications. Through this easy-to-use web portal, you have complete control over a range of available features including:

- Calling Number Delivery or Blocking
- Outgoing Call Blocking
- Voicemail
- Speed Dial
- Call Forwarding Unconditional, Busy, No Answer, Selective
- Do Not Disturb
- Anonymous Call Rejection
- Selective Call Acceptance
- Simultaneous Ringing
- Selective Call Rejection
- Find-Me-Follow-Me



AVAILABLE FEATURES

Incoming Call Manager

Auto Attendant with Dial by Name Directory

SimRing

Hosted Voice Mail

DIDs

SIP Trunking

Easy Migration to Hosted PBX

BENEFIT TO YOU

Each individual can decide where different types of calls should be sent.

Ensures effective call handling and projects a professional image. Adjust based on time-of-day, or enable during closed hours only.

Never miss an important call. With SimRing when your desk phone rings other lines – such as mobile or home phones – will also ring.

Ensure business continuity in case of power outage with backup voicemail service.

Direct Inward Dial extensions for your staff.

Session Internet Protocol (SIP) trunks are available to be used instead of traditional telephone digital trunks for connection to premises-based IP PBXs.

When you're ready to get a new phone system, you'll have a simple migration path to Hosted PBX Service.