

Combine your voice and data networks



What is SIP?

Session Initiation Protocol (SIP) works with Voice over IP (VoIP) to integrate voice and data on a single network so you can optimize the network capacity and flexibility of your existing PBX or key system investments. In other words, SIP makes your voice trunks speak the same language as your network resulting in faster, better quality and more flexible service. SIP Trunking with VOIP provides a new foundation for advanced services.

Why use SIP trunking



Consolidate
voice and data

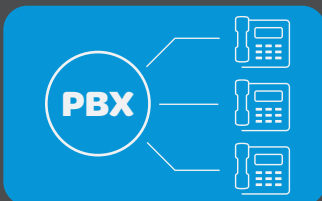


Improve productivity
and growth potential



SAVE
resources

SIP connects all your phone systems



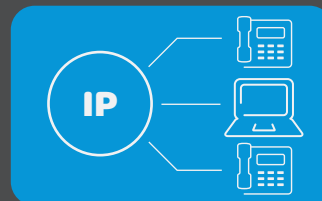
Private branch exchange

- Distribute calls as you like
- Less lines necessary compared to traditional trunks in many cases



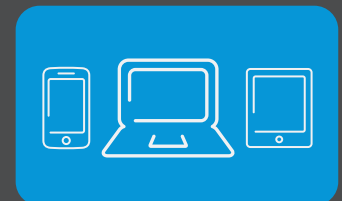
Telephones

- Helps scale your services as needed
- Keep what you have - no need to invest in new equipment for voice lines



IP phones

- Use your computer as a phone
- Use a hardware based IP phone that connects directly to your computer's network connection
- Seamless integration between IP phones and network



Mobile devices

- Stay connected from virtually anywhere
- Your phone is an extension of your office; make and receive calls over the network
- Shows your company caller ID on outbound calls

Contact AT&T for more information about SIP or to get a VoIP solution to meet your needs.

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