

Allied provides **SIP Trunking Services** designed to interface with your existing SIP-capable IP PBX equipment and enable connection to the Public Switched Telephone Network (PSTN).



MEETING Your Needs, Your Way

Allied can deliver SIP Trunking Services as a stand-alone solution or as part of a dynamic voice and data solution, utilizing a Dedicated Internet Access (DIA) connection to Allied's highly secure backbone network.

"ALWAYS ON" VOICE SERVICES

- Utilization of SIP Trunking enables voice communications and survivable design opportunities to ensure that your voice services are "always on" and available.
- Business continuity features include automatic timed failover.

EASE OF TRANSITION & SCALABILITY

- Ease of transition from traditional TDM networks: the end user with the current IP PBX environment does not change.
- Ability to add voice channels as needed according to business growth.

OPTIMIZATION & EFFICIENCY

- Optimal utilization of bandwidth by delivering both voice and data over the same private IP connection.
- Ability to leverage Wide Area Network (WAN) connectivity for voice call transport.*

**In a Bring Your Own Bandwidth solution (BYOB), Allied cannot control Quality of Service (QoS).*