Address more customers by allowing many more IP PBXes to be connected to your VoIP service!

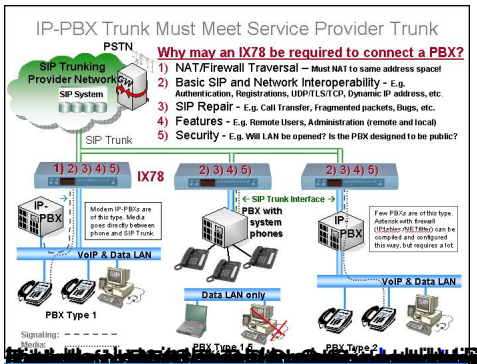
If you offer, or are planning to offer, SIP Trunking services, you need the Intertex IX78, a Router/Firewall/ADSL-modem with built in E-SBC (Enterprise Session Border Controller) functionality.

SIP Trunking – connecting business PBXs to operator's SIP based telephony infrastructure via IP instead of TDM lines – has opened up new market segments and new business opportunities for telecom operators. The shift to IP-PBXs at the customer site and to IP based infrastructure at the carrier side, is making SIP Trunking an obvious development. Today, almost every ISP, broadband provider or telco in general are planning to offer or are already offering SIP Trunking services.

However, getting business customers' telephony traffic, isn't just about having a good service and roll it out over a good IP pipe (dedicated/managed or over the Internet) – one also has to be able to connect the service to the customer's PBX in the customer's environment. Having a SIP Trunking interface on an IP-PBX is not sufficient!

Most obvious is the firewall/NAT traversal problem. Modern IP Phones and IP- PBXs reside on the business LAN which is protected by a firewall. But SIP communication does not pass ordinary firewalls or even NATs. A NAT/firewall traversal method is required, which is one of the functions the IX78 E-SBC provide.

In addition, there are many variants in the SIP capabilities and implementations in PBXs as well as the trunking service and delivery, something that the IX78 E-SBC can normalize. The picture below shows the functions and benefits the IX78 E-SCB brings when SIP Trunking various types of PBXs.





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