



What is SIP?

SIP, or Session Initiation Protocol, is a multimedia signalling standard; it allows you to communicate over IP connections. A SIP Trunk is the logical connection between your own telephone equipment and the service provider's SIP equipment. Telephone calls can be made and received over the SIP trunks, and these are commonly used as a replacement for legacy ISDN services.

Is SIP reliable?

We pride ourselves on our highly resilient network, with multiple redundant connections to multiple locations. The main threat to availability is the network connection between your premises and our data centre. By using connections provided and maintained by us, this threat is kept to a minimum.

How does SIP work?

Voice calls are routed over data networks using broadband, Ethernet, FTTC or similar connections. These calls then break out through the SIPLink Session Border Controllers (SBCs) into the public telephone network (PTSN).

Can I use my existing telephone system (PBX)?

In most cases, yes. Most modern telephone systems now support SIP, which means that you can connect directly to our service without third party equipment. If you have a telephone system that cannot support SIP, we are able to provide a gateway solution to convert from ISDN, enabling you to utilise SIP with your legacy system. We also have an ongoing programme of interoperability testing with various telephone system manufacturers. This ensures there are no incompatibilities between the PBX and SIP.

Is SIP cost effective?

Yes, compared to traditional ISDN solutions, SIP can save you up to 55% on your connection and rental costs. We can also offer significantly cheaper call charges compare to other services and providers.

Can I connect to SIP using my existing internet connection?

Yes, we allow connections via the internet, so you can use any third-party connection with SIP. For optimum reliability we would recommend using our QoS-enabled Broadband, Ethernet or FTTC products.

Can we keep our existing numbers?

Yes, we can port in numbers from almost any UK-based provider into the SIPLink platform.

Can I test the service before placing an order?

Yes. We can provide limited test accounts for you to use before committing to a contract.

How do I port my numbers?

Node4 manage the porting process on your behalf. We ask you to complete the relevant authorisation forms and we will then work with your current provider to migrate the numbers over to SIPLink. Number porting is Ofcom regulated and is subject to industry-standard processes and lead times.

What CODECs do you support?

We support G.711, G.729, G.723 AND G.722

Is there a charge to port numbers?

There is a small fee to port numbers in SIPLink. Charges depend on the type and quantity required.

How many simultaneous calls can I have?

It depends on the amount of bandwidth you have available, the CODEC used and the type of network connection. We can provide network connections with any amount of bandwidth that you require. As an example, a typical G.711 call will use between 80 and 100kbps.

Can you provide new telephone numbers?

Yes. We can supply telephone numbers from any UK Geographic location, regardless of your actual, physical location. Node4 can supply international, freephone and non-geographic numbers.

Does Node4 provide call bundles?

Yes, we offer a range of call bundles which include UK geographic and mobile minute, based on a contract period. Please contact your Node4 representative for details.