

Important & Helpful Information

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Voice Network Redundancy

DC Tech operates what we refer to as an Active/Active Geo-Redundant Network.

The concept is simple... spread infrastructure across multiple geographically dispersed locations and keep it all in sync. The execution however, is what's borderline magic :).

This method of redundancy helps ensure that if something goes wrong in one data center (think power loss, hardware failure, connectivity loss, etc...), services can keep running smoothly from another.

And because everything is always in sync, all your call routes, settings, and history are always available and working as expected.

This helps DC Tech & their Partners limit service disruptions for their customers, ultimately leading to a better customer experience.

How it works

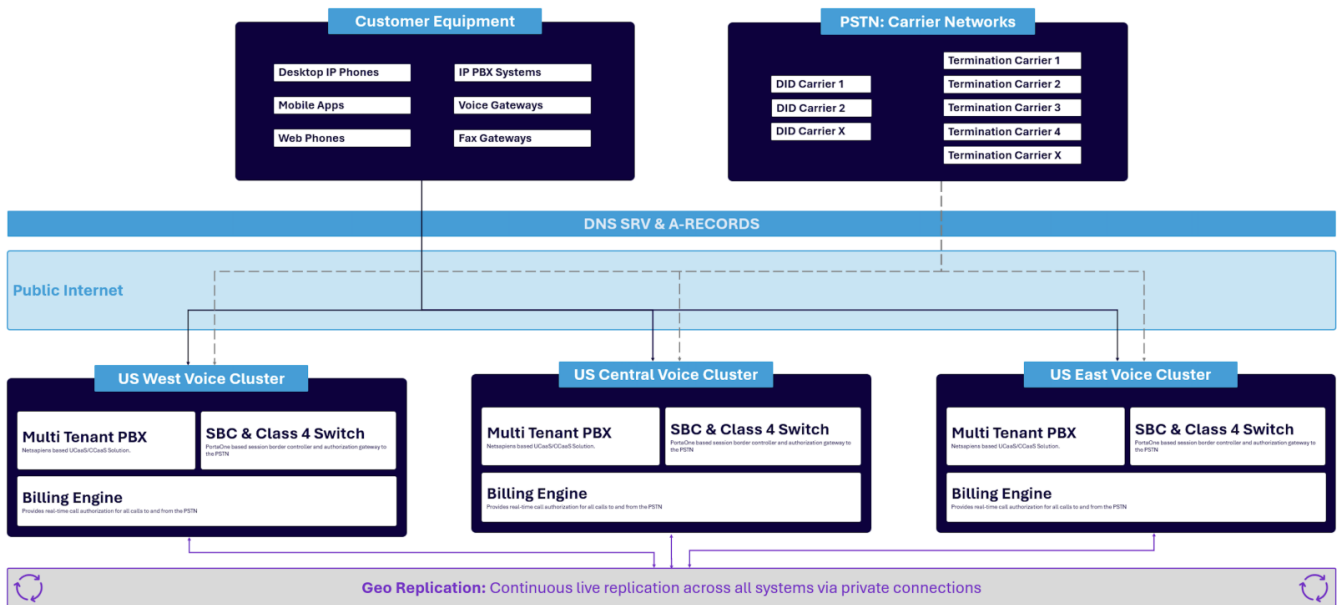
Each data center operates dedicated clusters of servers acting as standalone systems and as a backup to other data centers in the ecosystem.

These clusters are for critical components of the voice infrastructure like the session border controller, multi-tenant PBX, billing engine and nearly all other dependencies.

In most cases these clusters are horizontally scalable to increase processing capacity, and in some cases, managed by a load balancing system further running in High Availability (HA) locally.

Connectivity to a specific data center is determined by the DNS record used by the customers equipment. And aside from the replication magic mentioned earlier, DNS is the primary tool in keeping your customers service running.

When a customer is onboarded, its devices connect to their primary datacenter and server. Through out the course of the service use, all the settings and call history are both stored locally and actively replicated in real-time to all the other datacenters via our replication messaging bus.



Customer Endpoint DNS Redundancy Methods

SRV-Record: This is the default method implemented when auto provisioning your customers devices on our hosted PBX solution.

During bootup the device will query a DNS server for the SRV record of its configured SIP Server. Unlike a standard A-Record lookup, which contains a single IP, SRV responses will contain a list of servers with additional priority and distribution parameters.

That single query provides the device with several geo distributed datacenters it can connect to, with no need to reboot or flush local DNS settings.

Should a SIP Server become unresponsive or unavailable, during a re-registration or outbound call attempt, the device will then attempt to connect with the next server in its SRV list.

*For manually configured devices like IP PBXs, ATAs, Gateways, and the occasional one-off IP Phone, please check your devices administrative guides for instructions on how to implement SRV redundancy.

Standard A-Record: This method is inferior to SRV based redundancy and should only be considered as a last resort.

During bootup the device will query a DNS server for the A-Record of its configured SIP Server. The response will contain a single IP.

The device will then attempt to connect to the SIP Server via that IP.

Should that SIP Server become unresponsive or unavailable the device connection and its calls

will fail.

In the event a SIP Server is globally unavailable, and our Infrastructure Team has determined that it cannot be immediately restored then its corresponding A-Record will be updated with the IP of another SIP Server capable of handling its load.

*Note the TTL on our A-Records are 5 minutes. However, we cannot guarantee when global DNS servers will propagate and serve the changes. Further, customer devices must be rebooted so they can query and retrieve the updated A-Record

Dual Registration A-Records: This method is inferior to SRV based redundancy and is known to have issues with certain device <> firmware combinations.

In this method 2 SIP Servers are configured on the device.

How and when devices perform DNS lookups may vary from model to model. In most cases it will look like one of these examples.

During bootup the device will query a DNS server for the A-Record of its configured Primary SIP Server. The response will contain a single IP. In some cases it may query the A-Record for both the Primary and Secondary SIP Servers at the same time.

The device will then attempt to connect to the Primary SIP Server via that IP.

Should that Primary SIP Server become unresponsive or unavailable, and if the device hasn't done so already, the device will lookup and attempt to connect to its configured Secondary SIP Server.

*The switching experience from primary to secondary sip server varies from device to device.

Voice Network Details

DC Tech operates as a Active/Active Geo Redundant Infrastructure to best meet the needs of our partners and your customers.

Below are the addresses and related IPs for different segments of the network.

We recommend that you always use auto provisioning, and if manually configuring a device, use its SRV feature, to benefit from the best available redundancy methods.

We also suggest that you avoid using a specific DNS A Record or direct IP if manually configuring a device.

SIP Servers

IMPORTANT: If using this list to white list IPs on your local router or firewall, please ensure you whitelist all IP's or you may experience a service interruption during a maintenance or failover event.

	Address	DNS A record resolution IP	SRV record resolution IP
SBC Global			
SIP Trunking T38 IP-Fax Legacy Hosted PBX	sbc.simplelogin.net	192.92.8.30 Primary 64.52.82.30 Secondary	192.92.8.30 64.52.82.30
PBX East			
Hosted PBX Hosted Call Center	pbx1.mia.simplelogin.net pbx2-mia.simplelogin.net	192.92.8.71	All PBX IPs in varying order based on network needs
PBX Central			
Hosted PBX Hosted Call Center	pbx2.dfw.simplelogin.net	64.52.82.25	All PBX IPs in varying order based on network needs
PBX West			
Hosted PBX Hosted Call Center	pb1.sjc.simplelogin.net	64.52.83.20	All PBX IPs in varying order based on network needs

Codecs

DC Tech operates transparent proxies and does not perform transcoding. This ensure your media does not suffer any quality degradation from the transcoding process. Offered codecs will be negotiated by the SIP Server and a matching preferred codec will be used.

Supported Audio Codecs:

G.711U (PCMU)*

G.711A (PCMA)

G.722

G.729*

***Note:** These codecs should always be enabled to guarantee call completion.

Supported Video Codes:

H.263

H.264

DTMF

RFC2833 should always be the preferred DTMF method.

IP Fax

DC Tech supports IP Faxing via T.38 and G.711 fallback.

T.38 faxing is supported on most DIDs and Termination carriers and initiated via a T.38 re-INVITE.