ictalk3 Common Questions / FAQs

Common Questions

For all those frequently asked questions around ictalk3 support and provisioning.

- Do I have to dial 9 for external line?
- Experiencing one way audio
- Is it possible to have a group mailbox
- How do I access my voicemail
- What numbers are available
- Calls drop after 15 minutes
- Changing music on hold

Do I have to dial 9 for an external line?

It is common for organisations with existing phone systems to dial 9 before entering an external phone number.

With ictalk3 this is not required, the system intelligently works out if the call is to any of the following;

- Local ictalk3 extension
- ictalk3 extension under the company on another site
- Local national call outside of ictalk3
- National call in a different area code
- International call

Any calls made between ictalk3 extensions is not chargeable whilst all other calls are chargeable.

One way audio

This is where either party can hear the conversation but the other cannot, typically experienced on new installs or after an upgrade of network hardware.

SIP ALG

The most common cause is SIP ALG or SIP Session Helper which needs to be disabled for ictalk3 to operate.

Dropping packets

The other likely cause is dropping packets, with the fact VoIP operates with UDP error correction is not available as with TCP and so the packet is not made up, and subsequently audio is dropeped.

The most likely cause for this is the provisioning of ictalk3 over a dsl connection which proves to be flacky, or local interferance around the terminating hardware or faulty hardware. To combat these issues the cause needs to be identified;

For a brief checklist look below,

- Ensure the router is plugged into the master socket;
- Ensure all sockets are filtered, possibly replace;
- Run a continous ping to a destination on the Internet (such as your favourite website)

ping <destination> -t

Hold control and press c to cancel.

- Run a tracert to the same destination to see where packets are dropped
- Raise a ticket with Tech Support to run a line test on the connection

See our article on Wireshark tracing for fault finding (https://kb.ic.uk/article.php?id=237) for more in-depth troubleshooting

I require a group mailbox is this possible?

There is not feature a feature on ictalk3, however there are a couple of scenerarios to support this;

Option 1

Configure another user without a phone on the premium package, then set the user to DND. Calls can be routed to this user where voicemail messages are left. Voicemail can then be retrieved by any user whom knows the dummy voicemail users extension and pin number.

Option 2

Another method depending on the users role and potential confidential material is to route all voicemail to an existing user and provide access details to all required users.

How do I access my voicemail?

For those users whom have voicemail enabled there are a number of accessing your voicemail:

Local Extension

It is possible to access the voice mail system from the local handset itself.

Messages

On some handsets there are phones which have a messages button, which on pressing, dials the message centre requesting a pin number.

Feature code

On every handset supplied on the ictalk3 platform the user can dial the local extension of the handset and the prompt will request the appropriate pin number.

Voicemail Portal

For those who are trying to access the voice mail external to the site or from a handset not linked to the ictalk3 system a external telephone number set for this specific purpose must be dialled.

When this number is dialled the user is requested to enter the extension number and then the appropriate pin number to gain access.

Once access is granted you can interact with the voice mail system as if you were local and on the system.

What numbers can I buy?

ictalk3 allows you to port existing numbers into the platform but also to purchase new numbers (requirement of each site). Purchasing of 01,02 and 03 numbers are possible subject to availability, certain numbers like 0207 & 0208 are now exhausted and so not available. For full details please contact us.

Calls drop after 15 minutes

Once a call is established to ictalk3, if that call remains connected the platform will send a session audit message after 15 minutes. The phone should receive this and send acknowledgement back.

If however the port which the phones connection was assigned has changed there will be no response sent back to ictalk3, or any response received by ictalk3 will be ignored as the port has changed.

This is usually down to firewall/switch configuration such as NAT configuration and changing the NAT refresh timer to 120 seconds. Always worth ensuring SIP ALG or anything affecting the data packet is disabled.

Changing music on hold

If your site allows you to change our music on hold (license pack dependent) you will need the music on hold recording in the following format:

CCITT u-Law WAV with 8kHz, 8Bit, Mono attributes