

# VivoMeetings SIP Gateway

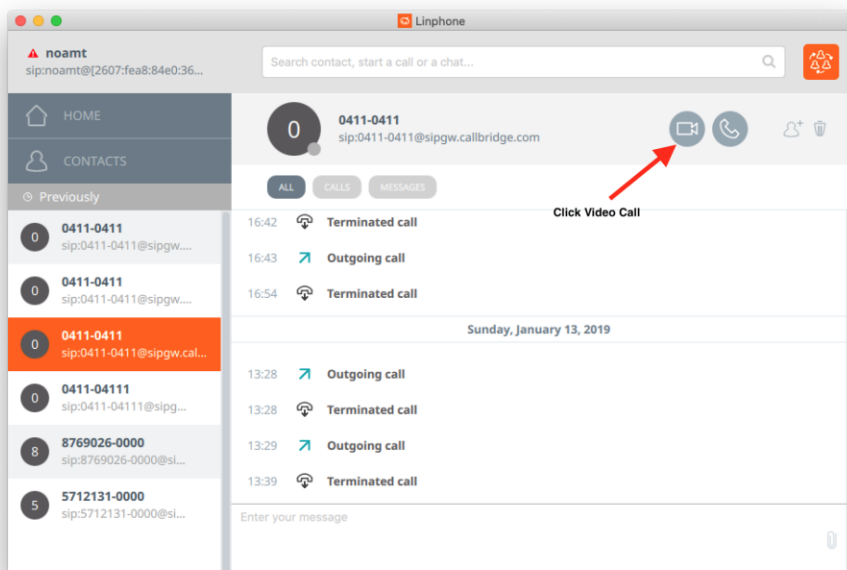
The SIP Gateway allows SIP based meeting room systems to join a VivoMeetings meeting room and participate fully in the video conference.

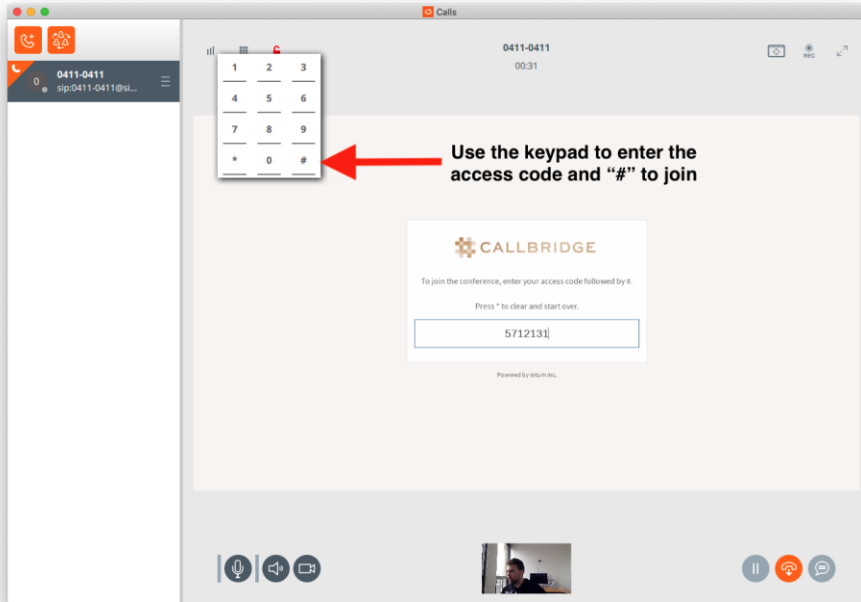
## Testing Clients:

- See the list of supported and tested client end points here - [https://docs.google.com/spreadsheets/d/1SdrDeQ8fSDgDCfgzV2o6bEQ3dHPYxlJIV7c\\_rdLOQYk/edit?usp=sharing](https://docs.google.com/spreadsheets/d/1SdrDeQ8fSDgDCfgzV2o6bEQ3dHPYxlJIV7c_rdLOQYk/edit?usp=sharing)
- You can also Test from any desktop/laptop computer using Linphone. Download Linphone here - <https://www.linphone.org/>

## Testing Steps:

- Get your access code from your account dashboard
- Join directly to a conference: <access-code>-0000@siggw.<domain.ext>:5060
  - Eg. [1234567-0000@siggw.vivomeetings.com:5060](https://1234567-0000@siggw.vivomeetings.com:5060)
  - \*Use your White Label domain in place of vivomeetings.com when testing one of your meeting rooms
- Join entering the access code **using the keypad** and **click “#”** to join
  - In the SIP client call 0411-0411 @siggw.<domain.ext>:5060





- Have other participants join the call using the same access code from their web browser, dial-in by phone, or with SIP connection

### Testing Plan:

- Verify SIP system automatically connects with audio and video and you can hear and see other participants and they can hear and see you
- Can mute and unmute your line
- Can hear all prompts normally (eg. mute, unmute)
- Are muted or unmuted based on conference mute mode
- Can see other participants video and screen sharing

### Current Known Issues:

- Maximum of 4 simultaneous SIP sessions currently are allowed
- Hand raise / lower is not yet implemented
- "-0000" part of the SIP URL is used for conference security codes, for now all security codes are bypassed
- One time access codes are not yet implemented

**NOTE:** Minimum supported Chrome browser is version 72