

 **SPARKLE** | Conferencing Suite
AUDIO





AUDIO

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THE EVOLUTION OF THE AUDIO CONFERENCE

Main features of the service

Conference Suite Audio is an innovative and unique audio conferencing service that allows the integration between traditional telephone conferences and those made in VoIP (Voice over IP). It is the first service to be connected via iNum, the new single global telephone number of the ITU - International Telecommunication Union. In any part of the world the same iNum can be used to participate in conferences without having to consult the PoP list. The possibility of connecting to VoIP also eliminates the cost of telephone traffic for those who use a softphone

from a PC, tablet or smartphone. Conference Suite Audio is not just audio conferencing: it offers real-time conference management with a few simple clicks and innovative management of reports and registrations.

You can also participate in conferences from tablets and smartphones. With the client integrated in the Moderator application it is possible to connect in VoIP with HD audio quality. Conference Suite Audio is also accessible by Skype.

Access telephone numbers (PoP)

Telephone accesses are available in over 40 countries worldwide with the possibility of activating up to 3000 PoP. In Italy, 6 expandable accesses are available on request up to 238. The Conference Suite Audio service is available 24/7/365 with no reservation required. You connect to the conference simply by calling the nearest access PoP, often a national number or a toll-free number. Local telephone operators will only bill a national call instead of an international call.



i Num, a unique number valid worldwide

In addition to traditional telephone PoPs, iNum is available: a new type of international number that makes access to audio conferences even easier and more immediate. iNum is based on the use of the +883-prefix created by the ITU and valid worldwide. iNum transports the voice in high definition, thus offering an audio quality that is considerably higher than traditional telephone.

Free app for IOS & Android



For those who are often traveling or attending audio conferences when out of the office, the Conference Suite Audio Call app allows you to access conferences with a simple click, both as an organizer and as a participant. It is the perfect app to manage conference calls from mobile IOS and Android: the preferred codes and phone numbers are always at hand and you no

longer need to consult the PoP list because the device will automatically locate the nearest PoP. By enabling VoIP you can use a third-party app to get further savings on telephone traffic. The app is free and can be downloaded from.



Conference Call from Skype

To join a Conference Suite Audio conference call from Skype (also with a smartphone and tablet) just add joinaudio.joinconferencing.com to your Skype contact list and then make the normal Skype audio call.

Virtual rooms for audioconferencing meetings

When subscribing to the service, the customer is assigned a virtual meeting room characterized by an access code for the moderator and one for the participants. To open the conference, the moderator must dial the most convenient access phone number and enter the room code. Participants will also have to dial a phone number from those available and enter the participant code that was previously provided by the moderator. It is possible to create rooms to be used only once for specific conferences: this option is useful in case codes are requested to be changed for each conference. Reservations for these rooms can be made independently via the SPARKLE service portal.

Spot conference calls

The Conference Suite Audio Spots are conferences for which the moderator and participant codes are valid only within the time frame of the reservation. These conferences can be created directly from the SPARKLE portal. To create a spot conference, access the SPARKLE portal (with your username / password) and enter the name of the conference, date and time of start and end. Once the conference is created, two emails are sent with information, access numbers and conference codes: one email is for the moderator and the other for the participants.

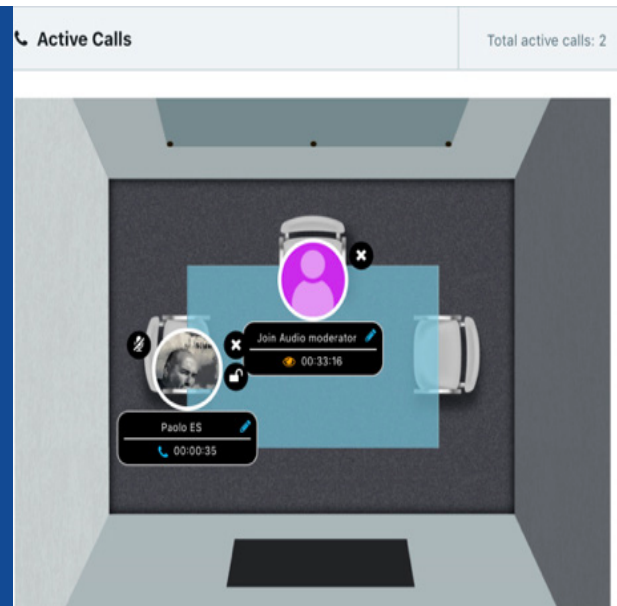
One-click audio conference control

Thanks to the Moderator application it is possible to easily control and manage audio conferences from a PC or a tablet connected to the Internet: see who is connected, control the microphones, activate and manage a Q&A session and much more. During the period of validity of the codes, the moderator can use the Moderator

application to manage the conference through a web interface. Thanks to the Moderator it is possible to directly control and manage audio conferences from a PC or tablet connected to the Internet: see who is connected, control the microphones, activate recording, download the audio file in MP3 format and much more.

FOR EXAMPLE IT IS POSSIBLE:

- See the numbers of the participants
- Open and close the microphones
- Manage Questions & Answers sessions
- Block access to the conference
- Call participants dial-out
- Participate in the conference in VoIP thanks to the integrated client



Conference management via the web is possible both for Conference Suite Audio Spots and for permanent conferences. Reset telephone costs with the VoIP client integrated in the Moderator.

Thanks to the client integrated in the Moderator application it is possible to connect in VoIP with high definition audio quality. Just one click to access the conference without using the phone. The numeric keypad also allows the insertion of DTMF commands. If you have a VoIP switchboard you can connect it directly to the Conference Suite Audio service. The platform that delivers the service is based on a VoIP core completely in SIP standard. This makes it possible to configure any switchboard to make calls to the Conference Suite Audio service directly in SIP, thus avoiding the costs of traditional telephone traffic.

24-hour operator assistance

The Conference Suite Audio automatic conferences are very simple: just dial the phone number of the nearest PoP and enter the participant or moderator code. Anyone who prefers to be connected to the conference by an operator can count on a call center available 24/7 and multilingual: Italian, English, German, French, Spanish and Russian.



SUMMARY OF THE SERVICE

- No hardware or software to install on company servers
- More than 40 PoPs in as many countries
- Activation on request of new PoPs, with more than 3000 numbers available
- Access also via iNum, the only worldwide phone number
- Softphone with HD audio integrated in the Moderator interface
- Compatibility with standard SIP phones and softphones, even on smartphones and tablets
- Compatibility with any hands-free audio conference phone
- Conferences with all telephone providers
- Possibility of SIP trunking with company IP switchboards
- Narrow band audio codec: G711, G729, iLBC
- HD audio codec: Wideband G722, G722-1, G722.s (AMR-WB), ISAC, SILK
- Conference control via Moderator interface or via DTMF
- Recording conferences via DTMF
- Easy call flow: a code for the moderator and a code for the participants
- Call flow modifiable from the service web portal
- Permanent audio rooms with up to 100 participants, expandable on request
- Spot conferences with timed access codes
- Conferences with operator concierge
- Dial out via DTMF
- Operator assistance in Italian, English, French, German, Spanish, Russian
- Detailed reports for cost centers, for users and for conferences



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