



### e:Presence

# **Teleconference Service**

## **Instructions for using H.323/SIP terminals**

Version 4.0

October 2024

#### INTRODUCTION

e:Presence teleconference service is mainly addressed to the academic and research community of Greece, allowing its members to organize and carry out high-definition teleconferences, reducing the relevant costs of travel and increasing productivity. The teleconferences through the e:Presence service have now replaced the vast majority of meetings which traditionally were held through the required physical presence of the participants, such as Foundation Councils, Meetings for the development of faculty members, Meetings for research and development programs, Committees for the examination of postgraduate theses or doctoral dissertations.

e:Presence teleconference service is based on Zoom's video conferencing technology. It offers a handy management environment with increased capabilities and implements an extremely easy process for the end users, in order to access directly the video conference meeting. Here, we will cover all the necessary instructions for those that use an H.323/SIP terminal in order to participate in a teleconference.

- Up to 1000 participants allowed in a teleconference
- Duration of a teleconference up to 24 hours

#### ΧΡΗΣΗ Η.323/SIP ΤΕΡΜΑΤΙΚΩΝ

In case of using H.323/SIP device, the "Connect" button should be clicked on ("Conferences" tab when logged-in) when the conference has started.

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Conferences								
In order to participate in e:Presence confer Zoom Client for Meeting. After you finish the i zoom client window.	ence calls, you need to nstallation, please close	o install the the pop-up	Expe Wou	eriencing a co Id you like to	nnection pro rate our serv	blem? press rice? Click HE	HERE! RE!	
Current confe	rences			Futu	re confere	ences		
Test Time:12:15-1 Moderator:	2:45							
		_						

The call is initiated by Zoom Cloud to the terminal of the participant; the IP address or SIP Uri are needed, in order to ensure that <u>only</u> the invited user participates in the teleconference. When the corresponding field is filled the green button "Dial" should be pressed.

	Connect to conference					
at st	Please enter the IF the conference, an name can only con	P address or the SIP URI of the system that you want to connect to id your name as you want it to appear within the conference. The tain english characters or numbers.	IE RE			
	IP/SIP URI: Enter IP address or SIP URI					
	Όνομα:	Nasos_Vaios_6				
	Make s When ye Answer	ure that the system is on and can accept incoming calls. ou click the Dial button, e:Presence will dial your system. the call to connect to the conference.				
		Dial <b>L</b>	Close			

The call should be answered within 30 seconds. If not, there is also the option to repeat the call by pressing the "Redial" button.

	Connect to conference				
If you are connected, you may close this dialog. If you are not connected, you may click the Redial button after 30 seconds.					
	00:24				
	Redial 📞 Clo	se			

*Important:* The device should be configured so that it may accept incoming calls. The necessary ports for incoming calls by Zoom Cloud are the following:

H.323: 1720 TCP port

SIP: 5060 TCP/UDP ports

<u>Note</u>: In order to establish a connection, depending on the terminal device used, more TCP/UDP ports may be needed (firewall settingsujj).

In case the connection cannot be established following the above instructions, there is the alternative to dial in the conference by clicking the "Dial Numbers" button and following the instructions depicted.

<u>Note:</u> In this case, the participant cannot be identified by the service.

Connect to conference			
If you are connected, you may close this dialog. If you are not connected, you may click the Redial button after 30 seconds. You may dial from your system into the conference if you click the following button. Dial numbers			
Redial 📞 Cl	ose		

	Connect to conference				
AT and to a of t	TENTION: If you dial into the conference from the conference report may contain only the conference participant. If you agree with that he following dial strings. If you do not agree w H.323: 213.19.144.110##61263133157	a your system, we cannot certify your identity IP address of your system, without relating it t, you may dial into the conference using one <i>i</i> th that, you may return to the previous step. SIP: 61263133157@213.19.144.110			
		Return 🗲 Clo	ose		