

Supported Telephony Functions

ProCall 6 Enterprise Windows Client since Version 6.4.0

PBX: bintec elmeg be.IP plus

PBX Version: R 10.2 Rev6 SP2

bintec LAN TAPI 64bit, Version 3.1.1.0

Action	TAPI	SIP
Dial	✓	
Hang up	√	 ✓
Answer Call	√	ď
Hold Call		ď
Call Back		<u> </u>
Reject Call	√	
Forward Call (without answering)	√ 1	√ ²
Forward Call (after answering)	0	 ✓
Setup Consultation Call	 ✓	₫
Toggle	√	₫
Connect Consultation Call	O ³	
Initiate Second Call (e.g. from Hold)	√	
Accept Second Call	√	 ✓
Reject Second Call	0	₫
Forward Second Call (without answering)	√	
Forward Second Call (after answering)	o ⁴	O ⁴
Create Conference Call (3 Participants) *		₫
Add further Participant to Conference Call	0	
Remove Participant from Conference Call	 ✓	
Pickup Call:		
TAPI - TAPI	√5	_
TAPI - SIP via Feature Code	√ 6	_
• SIP - TAPI	_	√
SIP - SIP via Feature Code	 ✓	<u> </u>
DTMF	 ✓	₫
Dial Ringing Tone	◊	_
<u>Call Forwarding</u>		
provided by PBX via TAPI		0
via Feature Code	<u> </u>	₫
 provided by UCServer⁷ 	_	V
Do Not Disturb (via Presence)	V	V
Do Not Disturb (via Line)	✓	√

	Function available
	Available/Function planned/Realization
_	Not relevant
0	Not applicable due to technical limitations
\wedge	Function not tested Full tests are pending

Known issues and features:

- 1 External calls only
- 2 Not from internal telephones
- 3 By replacing the handset
- 4 Function available but faulty: both stations are connected
- 5 No internal calls
- 6 Input on the telephone
- 7 Call diversion adjustable in the UCServer for: all/internal/external, always/on busy

ProCall 6.3 Enterprise

Mobile Apps

Supported Functions for SIP-Lines in Mobile Apps for connection via UCConnect:

Action	Android	ios
Dial	₫	V
Hang up	₫	V
Answer Call	₫	V
Reject (without answering)	 ✓	V
Second Call on SIP Line	Busy Line for Caller	Busy Line for Caller
Incoming Call on SIP Line while active mobile phone Call		
Answer call		☑ Mobile Call on hold
Reject		V
Incoming mobile phone Call while active SIP Line Call:		
Answer call	₫	 ✓
• Aliswei Call	SIP Line Call is terminated	SIP Line Call on hold
Reject	 ✓	V
DTMF	₫	V
Do Not Disturb (via Line)	√	√

^{*} Depending on the functions of the network provider

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