

Supported Telephony Functions

ProCall 6.2 Enterprise Windows Client V6.2.0

PBX:

Mitel MiVoice MX-One

Telefonanlage Version:

6.1 SP1 HF, 6.3 SP1

ECSTA for Mitel MiVoice MX-ONE

5.0.6.26

Action	TAPI	SIP
Dial	☑	☑
Hang up	☑	☑
Answer Call	☑	☑
Hold Call	☑	☑
Call Back	☑	☑
Reject Call	☑	☑
Forward Call (without answering)	☑	☑
Forward Call (after answering)	☑	☑
Setup Consultation Call	☑	☑
Toggle	☑	☑
Connect Consultation Call	☑	☑
Initiate Second Call (e.g. from Hold)	☑	☑
Accept Second Call	☑	☑
Reject Second Call	☑	☑
Forward Second Call (without answering)	☑	☑
Forward Second Call (after answering)	☑	○
Create Conference Call (3 Participants) *	☑	☑
Add further Participant to Conference Call	☑	☐
Remove Participant from Conference Call	☑	☐
<u>Pickup Call:</u>		
• TAPI - TAPI	☑	—
• TAPI - SIP via Feature Code	○	—
• SIP - TAPI	—	☑
• SIP - SIP via Feature Code	—	○
DTMF	☑	☑
Dial Ringing Tone	◇	◇
<u>Call Forwarding</u>		
• provided by PBX via TAPI	☑	○
• via Feature Code	☑	☑
• provided by UCServer*	—	☑
Do Not Disturb (via Presence)	○	☑
Do Not Disturb (via Line)	○	☑

*Call forwarding provided by UCServer adjustable for: internal/external/all, always/at busy

- ☑ Function available
- ☐ Available/Function planned/Realization
- Not relevant
- Not applicable due to technical limitations
- ◇ Function not tested. Full tests are pending

ProCall 6.2 Enterprise Mobile Apps V 6.2.0

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

Action	Android	iOS
Dial	✓	✓
Hang up	✓	✓
Answer Call	✓	✓
Reject (without answering)	✓	✓
Second Call on SIP Line	Busy Line for Caller	Busy Line for Caller
Incoming Call on SIP Line while active mobile phone Call		
<ul style="list-style-type: none"> • Answer call 	<input type="checkbox"/>	<input checked="" type="checkbox"/> Mobile Call on hold
<ul style="list-style-type: none"> • Reject 	<input type="checkbox"/>	<input checked="" type="checkbox"/>
Incoming mobile phone Call while active SIP Line Call:		
<ul style="list-style-type: none"> • Answer call 	<input checked="" type="checkbox"/> SIP Line Call is terminated	<input checked="" type="checkbox"/> SIP Line Call on hold
<ul style="list-style-type: none"> • Reject 	<input checked="" type="checkbox"/>	<input checked="" type="checkbox"/>
DTMF	<input checked="" type="checkbox"/>	<input checked="" type="checkbox"/>
Do Not Disturb (via Line)	<input checked="" type="checkbox"/>	<input checked="" type="checkbox"/>

* Depending on the functions of the network provider

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