



## Test Report

# **Yealink VCS Interoperability**

## **Test Report**

Dec 2015

Document Version 1.0

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## 1. Revision History

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<b>Version</b>	<b>Reason for Change</b>	<b>Date</b>
1.0	Draft version	Dec 2015



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## 1 Product Versions

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Access Device under Test: **VC400/VC120**  
Access Device Type: **VCS Phone**  
Access Device Version: **30. 20. 254. 22**  
Endpoint Software Version: **Zoem-europe  
RoomConnector/1.0.1(h323plus v1.24.0)**  
Completion Date : **2015/12/29**  
Completed By (Name): **HSZ**  
Completed By (Company): **Yealink**

## 2 Test Results

Complete the test report by marking an “X” in the appropriate column: **Pass, Fail, NA** (Not Applicable), **NS** (Not Supported), **NT** (Supported but Not Tested).

Test Case Title	Priority	Pass	Fail	NA	NS	NT
<b>1.Basic Functions</b>						
<b>1.1 Registration</b>						
1.1.1 DUTS – REGISTER ( <b>SIP</b> )	P0				X	
1.1.2 DUTS – REGISTER ( <b>H323</b> )	P0				X	
<b>1.2 Establish Call</b>						
<b>1.2.1 Establish Call_SIP</b>						
1.2.1.1_1 Audio Call through IP	P0				X	
1.2.1.1_2 Video Call through IP	P0	X				
1.2.1.1_3 Audio cornet call (registered SIP server's account)	P0			X		
1.2.1.1_4 Video cornet call (registered SIP server's account)	P0			X		
<b>1.2.2 Establish Call_H323</b>						
1.2.1.1_1 Audio Call through IP	P0	X				
1.2.1.1_2 Video Call through IP	P0	X				
1.2.1.1_3 Audio cornet call (registered GK)	P0			X		
1.2.1.1_4 Video cornet call (registered GK)	P0			X		
<b>1.3CodecNegotiation</b>						
G722		X				
PCMU		X				
PCMA		X				
G722.1c(48kb\s)					X	

G722.1c(32kb/s)					X	
G722.1c(24kb/s)					X	
G722.1(24kb/s)					X	
<b>1.4 DND</b>						
1.4.1 DND Before Call				X		
1.4.2 DND In Call	P0			X		
<b>1.5 Mute</b>	P0	X				
<b>1.6 DTMF</b>	P0	X				
<b>1.6.1 DTMF_SIP</b>						
1.6.1.1 Audio Call	P0				X	
1.6.1.2 Video Call		X				
<b>1.6.2 DTMF_H323</b>						
1.6.2.1 Audio Call		X				
1.6.2.2 Video Call		X				
<b>2. Advanced Function</b>						
<b>2.1 Encryption</b>						
2.1.1 SRTP(SIP)		X				
2.1.2 AES (H323)	P0	X				
<b>2.2 Presentation</b>	P0					
2.2.SIP(BFCP)					X	
2.2.3 H.323(H.239)	P0	X				
<b>2.3 Far-end Camera Control</b>	P0					
2.3.1 FECC(SIP)					X	
2.3.2 FECC(H.323)	P0				X	
<b>2.4 H.264 HP</b>	P0				X	
<b>2.5 H264 Base</b>	P0	X				
<b>2.6 H263</b>	P0				X	
<b>2.7 Flow Control</b>	P0					X

