

# VoIP 使用手册

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# 简介

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- ◆ 音视频通话数据统计检测

# 普通设置



# 普通设置

中国联通 14:56 97%

取消 +8616000000485 存储

帐号名 +8616000000485

用户名 +8616000000486

密码 ●●●●●●

鉴权名 460070000000486@ims.mnc...

昵称 +8616000000486

服务器地址 120.197.90.65

服务器域名 gd.ims.mnc007.mcc460.3g...

注册类型 CMCC >

注册无鉴权头

中国联通 14:55 96%

< 返回 高级

SIP

传输类型 TCP >

服务器端口 5260

语音

编解码 >

发送端增益控制

接收端增益控制

发送端自动降噪

接收端自动降噪

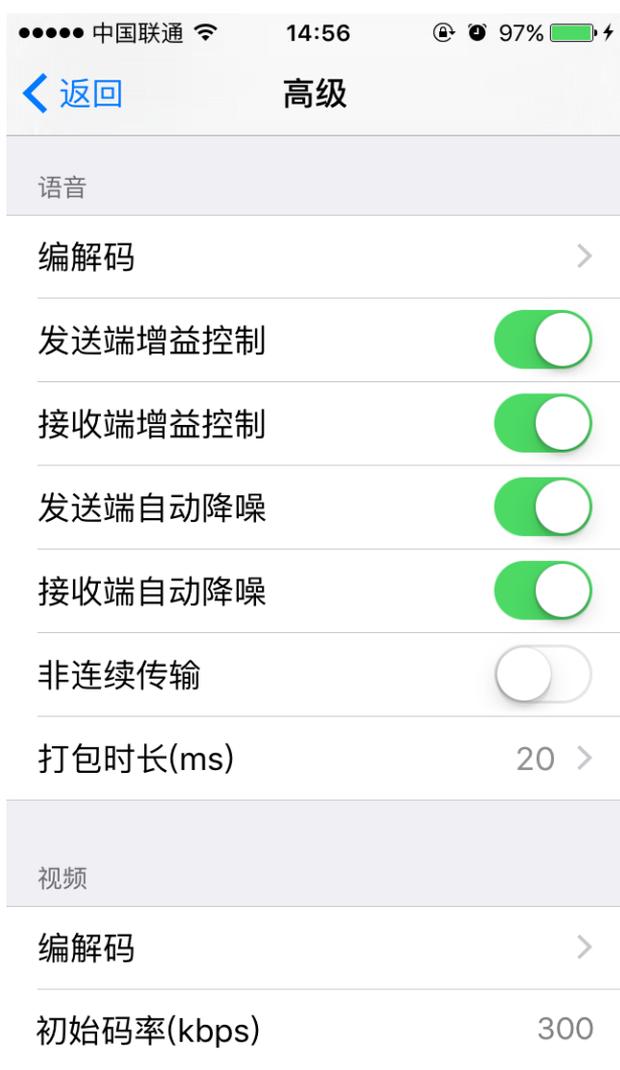
非连续传输

打包时长(ms) 20 >

设置好用户名、密码、鉴权名、服务器地址、服务器域名、传输类型、服务器端口

设置注册类型为VoIP

# 音视频参数设置



可设置编解码或其他参数

## Voice Parameters Intro

- Local Port, Remote Port: set local and remote transmission ports
- RTCP Mux: RTCP Port multiplexing
- Codec: select voice codec
- Payload: RTP payload value (not recommended to change)
- Bit rate: set bit rate (not recommended to change)
- Packet Length: set packet length (not recommended to change)
- AEC: acoustic echo canceller, including normal, OS, adaptive-FDE, adaptive-SDE, fixed delay mode  
FDE: formant delay estimate    SDE: spectrum delay estimate
- ANS: auto noise suppression
- AGC: auto gain control
- VAD/DTX: voice activity detection and discrete transmission
- AGC Target(dB): AGC target value with -10dB by default

## Video Parameters Intro

- Local Port, Remote Port: set local and remote transmission ports
- RTCP Mux: RTCP port multiplexing
- Codec: select video codec
- Payload: RTP payload value (not recommended to change)
- Bit rate: set bit rate
- Frame Length: set video frame rate
- Resolution: select video resolution
- Camera: select camera(front/back/off)
- ARS: auto rate sensing to make bit rate adaptive to network
- Resolution control: auto control resolution according to network
- Framerate control: auto control frame rate according to network
- Save mode: save network band width mode

# Video Parameters Intro

## **Transmission enhancement:**

- Color Enhancement: close by default
- FEC/RED: forward error correction
- NACK: Negative Acknowledgement
- TMMBR: temporary maximum media stream bit rate request, need be used with ARS
- FIR: full intra request
- VP8 RPSI: reference picture selection for VP8 codec
- H.264 small nalu: small network abstract layer unit for H.264, which means a small nalu packet is less than a UDP packet
- Key Period: interval between two key frame, use values by default
- Render Mode: video render mode(surface/openGL ES 2.0)

# Settings

ACCOUNT

+8616000000489



Add Account... >

Confirm be

点击账号区域发起  
登录

Vibrate whi

Quick Dial >

Ringtone

Music Box >

UDP Run in Background



Feedback >

中国联通 17:49 97%

+8616000000489

+86 160000000486

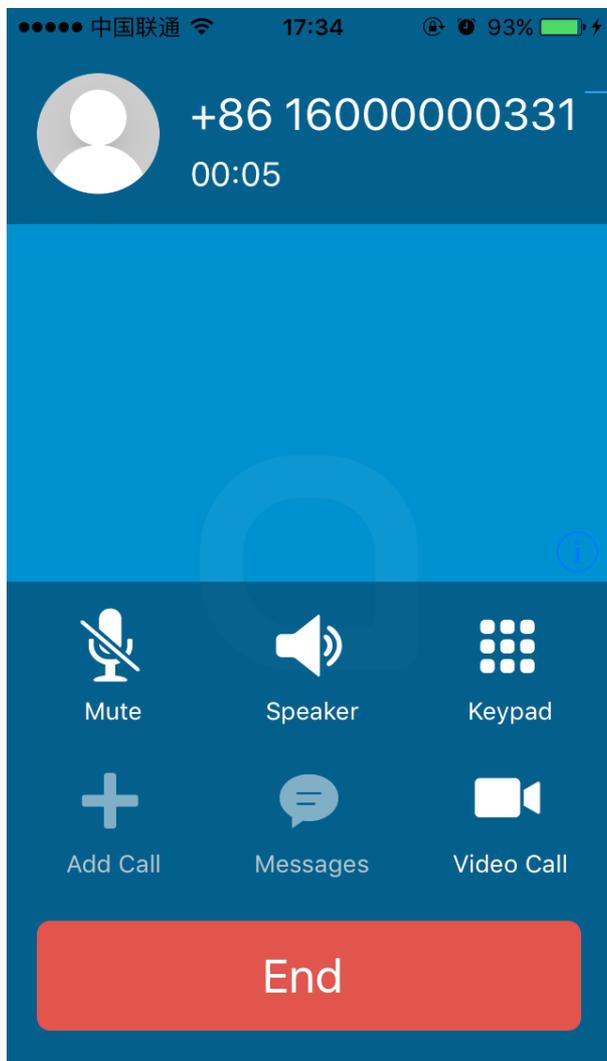
Add to Contacts



Messages Recents Contacts Keypad Settings

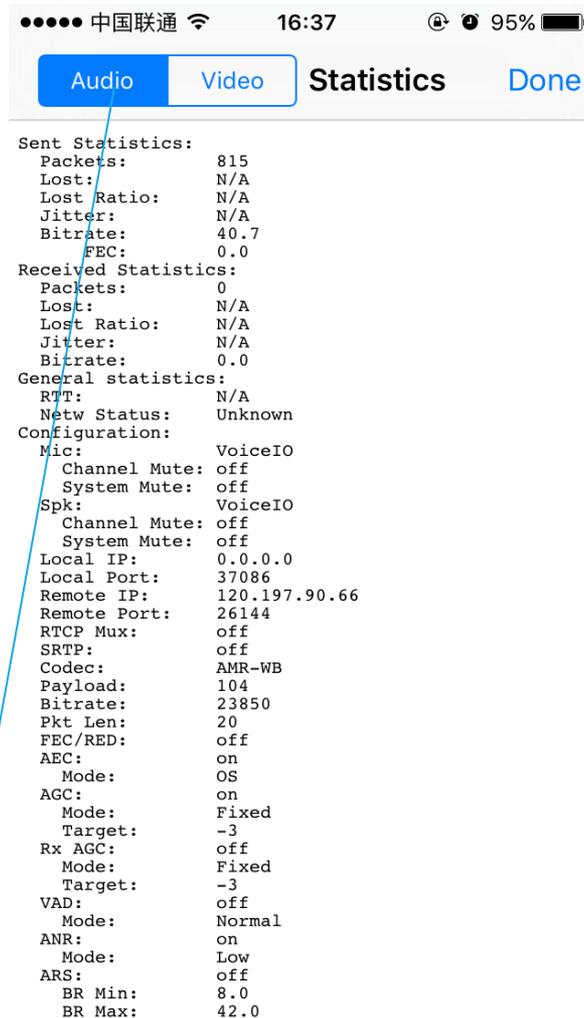
输入被叫号码，点击话筒按钮发起音频通话，点击摄像头按钮发起视频通话

# 音视频数据统计检测



点击该按钮查询通话数据，比如码率、丢包率、分辨率等

点击该按钮切换至语音数据说明界面



谢谢