

# WCDMA UGxx Audio Design Note

### **UMTS/HSPA Module Series**

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## **About the Document**

## **History**

Revision	Date	Author	Description
1.0	2015-9-7	Felix YIN/ Jun WU	Initial
1.1	2015-10-23	Felix YIN	<ol> <li>Updated PCM information in Chapter 2</li> <li>Updated <data_length> value for AT+QDAC</data_length></li> </ol>
1.2	2015-12-21	Felix YIN	Updated PCM timing in Chapter 2



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# 1 Introduction

This document mainly introduces how to use Pulse Code Modulation (PCM) digital interface with Quectel UGxx modules.



## 2 PCM Characteristic

Quectel UGxx modules provide one PCM digital interface for audio design, which supports the following features:

- Supports16, 32 bit mode with short frame synchronization, the PCM supports 32 bit mode by default.
   The default configuration of PCM codec is AT+QDAC=1,0,0,2,0,0. Refer to document [3] for more details.
- Supports master and slave mode.
- Supports audio sample rate 8 kHz.

The following table shows the pin definition of PCM and I2C interface.

**Table 1: Pin Definition** 

Pin Name	1/0	Description
PCM_IN	DI	PCM data input
PCM_OUT	DO	PCM data output
PCM_SYNC	DO	PCM data frame sync signal
PCM_CLK	DO	PCM data bit clock
I2C_SCL	OD	I2C serial clock
I2C_SDA	OD	I2C serial data

**Table 2: IO Characteristics** 

Parameter	Description	Min.	Max.	Unit
V <sub>IL</sub>	Low-level input voltage	-0.3	0.35	V
V <sub>IH</sub>	High-level input voltage	1.3	1.85	V
V <sub>OL</sub>	Low-level output voltage	0	0.25	V



V <sub>OH</sub> High-level output voltage 1.55 VDD_EXT V
--

The voltage of VDD\_EXT is 1.8V, please refer to **document [1]** for details.

In PCM audio format, the MSB of the channel included in the frame (PCM\_SYNC) is clocked on the second CLK rising edge after the PCM\_SYNC pulse rising edge. The period of the PCM\_SYNC signal (frame) lasts for Data word bit +1 clock pulses.

The firmware of module has integrated the configuration on NAU8814/ALC5616/MAX9860 application with I2C interface. **AT+QDAC** command can be used to configure the external codec chip linked with PCM interface, and refer to *document [3]* for more details. Data bit is 32 bit and the sampling rate is 8 KHz. The following figure shows the timing of the application with ALC5616 codec.

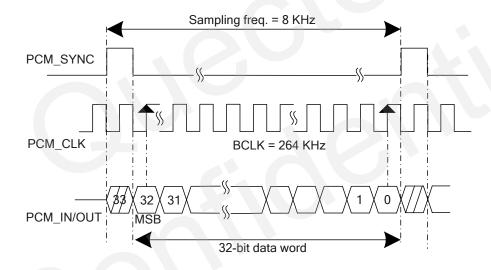


Figure 1: PCM Master Mode Timing



For the relations between peripherals, modules and PCM CODEC, please refer to the following figure.

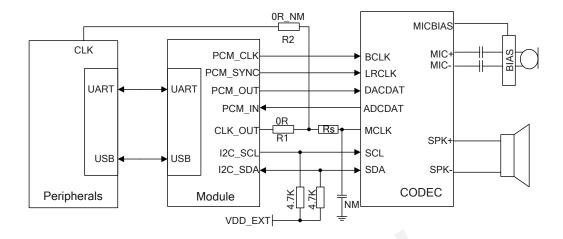


Figure 2: The Connection Diagram of Peripherals, Module and Codec

#### **NOTES**

- 1. The communication between the module and peripherals can be realized by UART or USB interface.
- 2. The MCLK of codec can be provided by peripherals or module. When the MCLK is provided by module, R1 is mounted and R2 is not mounted. When the MCLK is provided by peripheral, R2 is mounted and R1 is not mounted.
- 3. It is recommended to reserve RC (R=22 $\Omega$ , C=22pF) circuit on the PCM lines, especially for PCM\_CLK.
- 4. UGxx modules provide a digital clock output (CLK\_OUT) for an external audio codec, the CLK\_OUT function is disabled by default. When CLK\_OUT is required, AT command is used to provide the codec with a 13/26MHz clock generated from the module. A RC (e.g. R=22Ω, C=47pF) circuit is recommended to be reserved on CLK\_OUT line. Refer to *document [3]* for details. If unused, keep this pin open.

**AT+QPCMON** is used to enable PCM\_CLK (BCLK) & PCM\_SYNC when there is no audio application. The output frequency depends on the settings of **AT+QDAC**: sync=<**sample\_rate>**, PCM\_CLK(BCLK)=<**sample\_rate>** \* (**<data\_length>**+ 1).

AT+QDAC=1,0,0,2,0,0	
OK AT+QPCMON=1	//Enable PCM SYNC & BCLK, SYNC=8k, BCLK=264K
OK AT+QPCMON=0	//Disable PCM SYNC & BCLK
ок	



# 3 Audio Circuit Design

Codec ALC5616, NAU8814 and MAX9860 have been approved for PCM application.

Pin25 provides a clock ranges from 32k to 26M. This clock can be used to provide MCLK for codec. When ALC5616, NAU8814 or MAX9860 is using, Pin25 will output a 26M clock by default. **AT+QPCMCLK** is used to enable this clock.

AT+QCLKOUT=1, 2	//Output 26M frequency	
ОК		
AT+QCLKOUT=0	//Close output	
ОК		

The following chapters will show the audio design circuit with these codec.



## 3.1. PCM Codec Design Circuit with ALC5616

The following figure shows PCM application with ALC5616 from Realtek (<a href="http://www.realtek.com.tw/">http://www.realtek.com.tw/</a>), and I2C interface is equipped to configure codec.

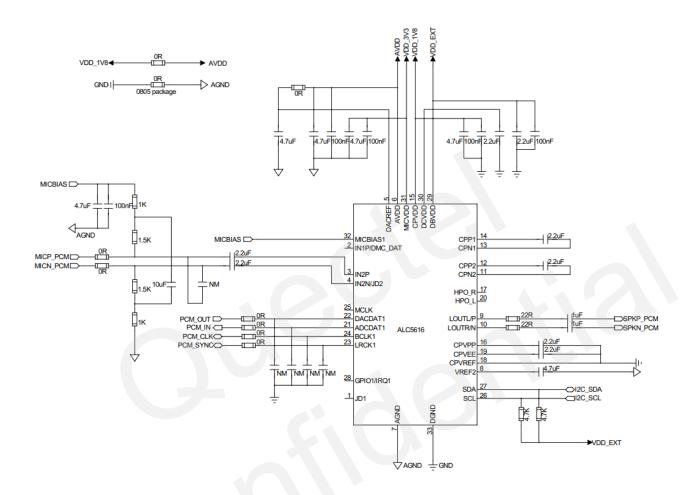


Figure 3: ALC5616 Codec Design

#### **NOTES**

- 1. The RC filter circuit (R=22ohm, C=22pF) needs to be installed on PCM\_CLK line.
- 2. VDD\_3V3 and VDD\_1V8 are needed to be supplied externally.
- 3. Set AT+QDAC=3 to choose ALC5616, MCLK is not needed, and ALC5616 uses internal PLL to generate system clock.



## 3.2. PCM Codec Design Circuit with NAU8814

The following figure shows PCM application with NAU8814 from Nuvoton (<a href="http://www.nuvoton.com">http://www.nuvoton.com</a>), and I2C interface is equipped to configure codec.

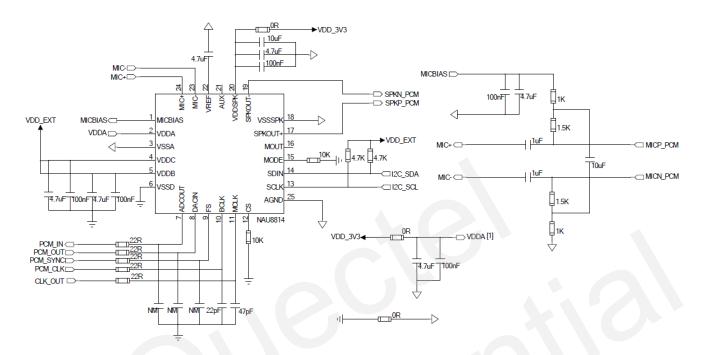


Figure 4: NAU8814 Codec Design

#### **NOTES**

- 1. The RC filter circuit (R=22ohm, C=22pF) needs to be installed on PCM\_CLK line.
- 2. VDD\_3V3 is needed to be supplied externally.
- 3. Set AT+QDAC=2 to choose NAU8814, please provide 2M external crystal for NAU8814. The CLOCK\_OUT pin can also provide 26M clock for NAU8814, but the quality of the voice is worse than 2M external crystal.



## 3.3. PCM Codec Design Circuit with MAX9860

The following figure shows PCM application with MAX9860 from Maxim (<a href="http://www.maxim-ic.com">http://www.maxim-ic.com</a>), and I2C interface is equipped to configure codec.

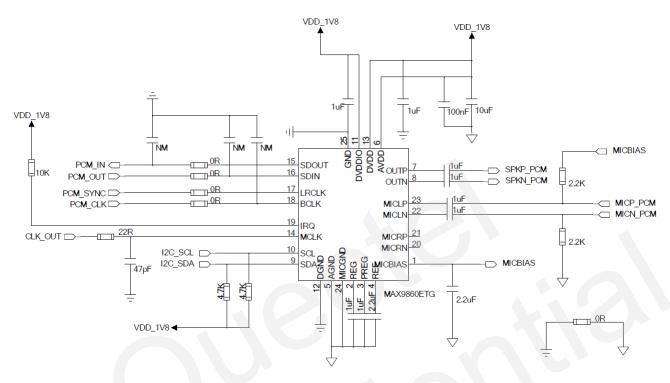


Figure 5: MAX9860 Codec Design

#### **NOTES**

- 1. The RC filter circuit (R=22ohm, C=22pF) needs to be installed on PCM\_CLK line.
- 2. Set AT+QDAC=4 to choose MAX9860, the module will provide 26M MCLK by default.



## 3.4. Microphone Interface Design

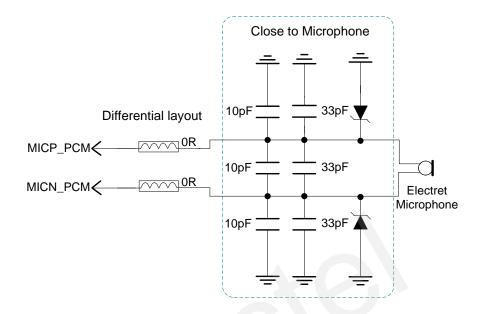


Figure 6: Microphone Design

## 3.5. Receiver Interface Design

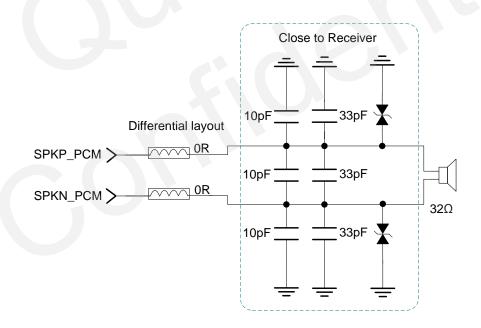


Figure 7: Receiver Design



## 3.6. Earphone Interface Design

The following figure shows the single-ended application for earphone design.

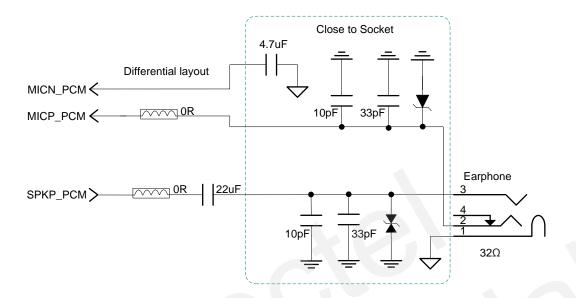


Figure 8: Earphone Design

## 3.7. Speaker Amplifier Design

If the  $8\Omega$  speaker is applied, it is recommended to add an audio amplifier.

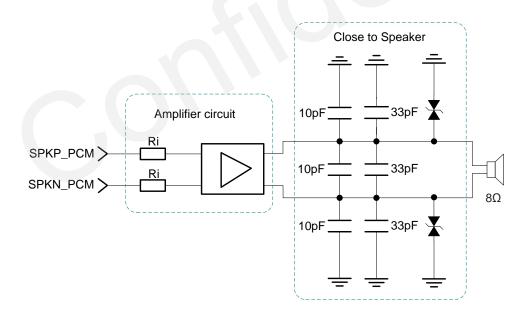


Figure 9: Speaker Amplifier Design



# 4 Design Consideration

## 4.1. Power Supply for PCM Codec

For different PCM codec, the supply voltage is different. It's better to supply power for PCM codec with dedicated LDO, and not to share this power with other circuits.

## 4.2. Audio Layout

Power supply ripple, unbalanced ground and RF burst radiation have negative effect on audio layout. Layout of MICP\_PCM/MICN\_PCM, SPKP\_PCM/SPKN\_PCM must meet the rule of differential signal. Moreover, these two pairs of signals should be separated from each other by flooding ground to avoid echo issue from SPK signal to MIC signal. Figure 10 shows an example.

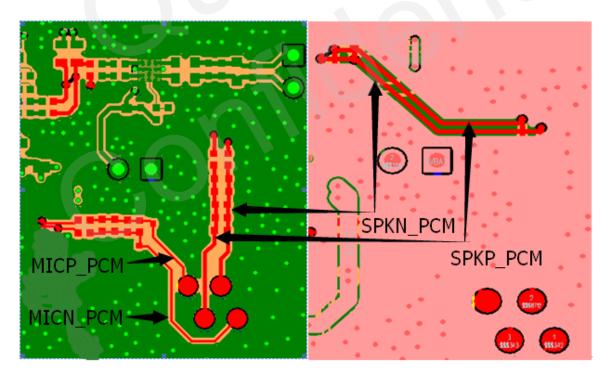


Figure 10: Audio Layout



#### 4.3. TDD Noise

It is important to avoid or reduce TDD noise in audio circuit design and layout, some suggestions are listed below:

- Different capacitors have their own self-resonant frequencies due to the different fabrication processes and materials. Generally, we choose 33pF capacitor (0603 package) to filter GSM900 interference and 10pF capacitor (0603 package) to filter DCS1800 interference on the power and analog audio signal lines. It is strongly recommended to add an electrets-microphone within two capacitors (10pF and 33pF) in handset and hands-free application. These two capacitors could largely suppress coupling TDD noise from RF interference.
- 2. Capacitors should be placed close to audio components or audio interface, and layout must be short.
- 3. Flood ground area should be as large as possible to reduce the ground impedance and improve ground integrality.
- 4. Reduce power voltage ripple, especially the power supply in audio circuits. A wide layout is used from power source (like adapter interface, battery connector, or LDO output pin) to audio power supply. Good antenna matching is also important to reduce power ripple.
- 5. The filtering capacitors and ESD protection devices should be connected to main digital ground. Separate analog ground and digital ground, and connect them at a single point on the PCB by one 0 ohm resistor to reduce digital interference and background noise.
- 6. Antenna must be stay away from audio components and layout of audio circuit. Antenna should be kept at least 5cm away from microphone.
- 7. Layout of power supply must be stay away from audio and cannot be parallel.

## 4.4. Mechanical Design

It is important to consider how to suppress echo in the equipment with hands-free function or in an application of which microphone and speaker are very close to each other.

The mechanical structure design has significant impact on echo issue. If it is not properly designed, the echo suppressing arithmetic in software could not make up for echo issue caused by bad mechanical structure, and even force to redesign.

Echo issue could be generated by several paths, as shown in Figure 11.



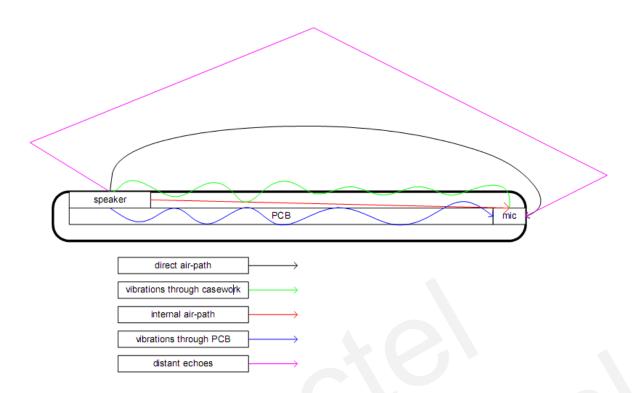


Figure 11: Five Echo Paths

In these five paths, internal air-path and direct air-path are the first influential factors. Other three factors (vibrations through casework, vibrations through PCB, distant echoes) are secondary.

To deal with echo issue from internal air-path, you can separate microphone from internal space of chassis by foam or rubber ring which can effectively suppress the inner echo interference. Figure 12 shows recommended design for microphone socket.

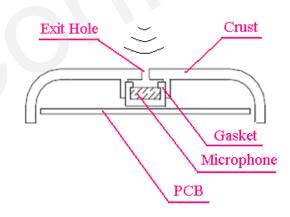


Figure 12: Microphone Socket Design



The best installation way of microphone socket is to encase microphone by silicone cover except for front cavity, and design a cylindrical hole whose center is the exit hole inside the chassis, make the microphone with silicone cover just fit the cylindrical hole, so as to only let voice enter into microphone from the exit hole, and not from leak of chassis interior. Certain air space room should be reserved in the front cavity of microphone as it is necessary for good microphone performance.

Figure 13 shows recommended design for speaker socket.

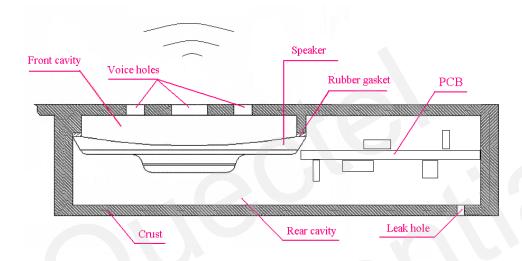


Figure 13: Speaker Socket Design

A good way to suppress the internal echo path is to seal the rear cavity of receiver, which is usually expensive. The rear cavity of receiver and speaker is important for good voice quality. A sealed rear cavity with sufficient space could produce a good voice output. An  $8\Omega$  speaker is often big and difficult to give an independently sealed rear cavity for it. However, sealing microphone socket in chassis is always useful. Furthermore, the whole chassis must be sealed as far as possible. If there is any unavoidable leak hole, keep it far away from microphone. If the leak hole is close to microphone, the voice coming from the hole could be picked up by MIC, and then leads to echo at the far end. If the leak hole is close to speaker, the output voice quality could be aggravated at certain extent.

## 4.5. Component of SPK

A SPK and receiver with higher sensitivity, flatter frequency response, less THD and impedance of  $32\Omega$  (receiver),  $16\Omega$  (receiver) or  $8\Omega$  (speaker) is recommended. These technical data are often shown in data sheet of SPK and receiver. For SPK, its frequency response and THD performance can be tested by speaker test system. SPK frequency and THD response are shown in Figure 14 and Figure 15.



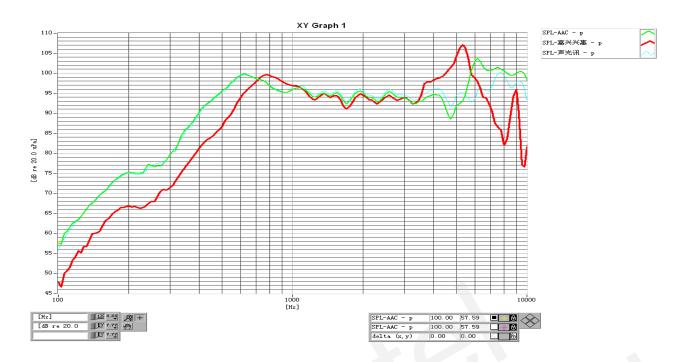


Figure 14: SPK Frequency Response

Horizontal axis is frequency, longitudinal axis is loudness and unit is dB.

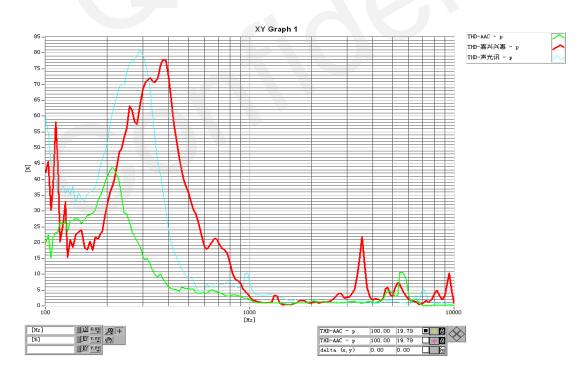


Figure 15: SPK THD



- 1. Horizontal axis is frequency, longitudinal axis is distortion and unit is %.
- 2. In Figure 15, the three colors represent three venders' products. After comprehensive comparison, the green one performs the best, and the blue one is worse, and the red one is the worst.

## 4.6. Components of Microphone

It is recommended to use an electret microphone with a sensitivity of -42±3dB/Pa @ 2V (not less than -44±3dB) and impedance of 2.2Kohm. If RF TDD noise is detected at the MIC, please contact MIC vendor for product with better RF suppression capability. Furthermore, the microphone channel circuit can be optimized to decrease TDD issue at MIC side.



# 5 Audio Tuning in Different Modes

There are three kinds of common modes: handset, headset, handfree. **AT+QAUDMOD** is used to choose modes. For each mode, there are some default settings, such as CLVL, CRSL, QSIDET, QMIC and QAUDGAIN.

If the audio is not good and AT+QAUDMOD has been executed, please follow the steps below to tune audio:

- Step 1: Execute AT+CLVL to tune downlink volume.
- Step 2: Execute **AT+QMIC** to tune uplink volume.
- Step 3: If the volume is a little smaller or higher and you have set QMIC & CLVL, please use AT+QAUDGAIN to tune it.
- Step 4: If you use the recommended codec, you can use AT+QAUDCFG to tune the codec volume.
- Step 5: Execute **AT+QSIDET** to tune side tone.
- Step 6: If there is echo or noise left, please use AT+QAUDCFG to tune EEC & NR.

### NOTE

Please do not set the volume too high, otherwise it will influence EEC & NR.

This following examples show how to tune audio in three different modes:

## 5.1. Speech Call in Handset Mode

The default mode is set to handset mode.

AT+QAUDMOD=0	// <mode>=0 is used for handset mode.</mode>
OK AT+CLVL=80	//Tune downlink volume.
OK AT+QMIC=0,12	//Tune uplink volume
ОК	



AT+QAUDGAIN=0,-1,50 //If downlink or uplink volume is a litter lower or higher,

please use AT+QAUDGAIN to tune it.

OK

AT+QSIDET=-100 //Tune side tone

OK

## 5.2. Speech Call in Headset Mode

Set AT+QAUDMOD=1 to use Headset mode.

AT+QAUDMOD=1 //<mode>=1 is used for headset mode.

OK

AT+CLVL=70 //Tune downlink volume.

OK

AT+QMIC=1,14 //Tune uplink volume

OK

AT+QAUDGAIN=1,60,-1 //If downlink or uplink volume is a litter lower or higher, please

use AT+QAUDGAIN to tune it.

OK

AT+QSIDET=-100 //Tune side tone

OK

## 5.3. Speech Call in Handfree Mode

Set AT+QAUDMOD=2 to use Headset mode. This mode is always used for vehicle-mounted devices.

AT+QAUDMOD=2 //<mode>=2 is used for handfree mode.

OK

AT+CLVL=50 //Tune downlink volume.

OK

AT+QMIC=2,12 //Tune uplink volume

OK



AT+QAUDGAIN=2,100,60 //If downlink or uplink volume is a litter lower or higher,

please use AT+QAUDGAIN to tune it.

OK

AT+QSIDET=-450 //Tune side tone

OK



# 6 Audio AT Command Description

This chapter will introduce the common audio related AT commands. For more details, please refer to **document [3]**.

## 6.1. AT+QDAC Digital Audio Interface Configuration

AT+QDAC is used to configure the digital audio interface. While <io>=1, you can define the PCM interface by yourself; while <io>=2, and the external codec chip linked with PCM interface is the NAU8814 model through the I2C, the module can be used directly and set by the default configurations. while <io>=3, and the external codec chip linked with PCM interface is the ALC5616 model through the I2C, the module can be used directly and set by the default configurations. while <io>=4, and the external codec chip linked with PCM interface is the MAX9860 model through the I2C, the module can be used directly and set by the default configurations.

AT+QDAC Digital Audio Interface	Configuration
Test Command AT+QDAC=?	Response +QDAC:(1-4),(0,1),(0-8),(0-2),(0,1),(0,1) OK
Read Command AT+QDAC?	Response +QDAC: <io>[,<mode>,<sample_rate>,<data_length>,<cha nnel_mode="">,<data_format>]  OK</data_format></cha></data_length></sample_rate></mode></io>
Write Command  AT+QDAC= <io>[,<mode>,<sample_ra te="">,<data_length>,<channel_mode>,&lt; data_format&gt;]</channel_mode></data_length></sample_ra></mode></io>	Response OK ERROR
Maximum Response Time	300ms



<io></io>	<u>1</u>	Digital PCM output (customer defined)
	2	Analog output (for default audio codec NAU8814)
	3	Analog output (for default audio codec ALC5616)
	4	Analog output (for default audio codec MAX9860)
<mode></mode>	<u>0</u>	Master mode
	1	Slave mode
<sample_rate></sample_rate>	<u>0</u>	8K
	1	11K
	2	12K
	3	16k
	4	22K
	5	24K
	6	32K
	7	44K
	8	48K
<data_length></data_length>	0	16-bit
	<u>2</u>	32-bit
<channel_mode< td=""><td>&gt; <u>0</u></td><td>Mono</td></channel_mode<>	> <u>0</u>	Mono
	1	Stereo
<data_format></data_format>	0	PCM
	1	I2S

#### **NOTES**

- 1. Configuration of **<io>** will be saved to NV immediately by default.
- 2. The module provides clock, and clock depends on <sample\_rate> and <data\_length>
- 3. When you select ALC5616, NAU8814 or MAX9860, please do not input other parameters, the PCM interface is set by the default configurations.

### **Example**

```
AT+QDAC=? //Query the range.
+QDAC: (1-4),(0,1),(0,8),(0-2),(0,1),(0,1)

OK
AT+QDAC? //Query current interface configuration.
+QDAC: 1,0,0,2,0,0

OK
AT+QDAC=1,0,0,2,0,0 //Set PCM interface to master mode, PCM format.

OK
AT+QDAC=2 //Select NAU8814.
```



OK

AT+QDAC=3 //Select ALC5616.

OK

AT+QDAC=4 //Select MAX9860.

OK

## 6.2. AT+QIIC IIC Read&Write

AT+QIIC is used to configure the codec via I2C interface.

AT+QIIC IIC Read&Write	
Test command	Response
AT+QIIC=?	+QIIC: (0,1),(0-FF),(0-FF),(1,2),(0-FFFF)
	ОК
Set command	Response
AT+QIIC= <rw>,<device>,<addr>,<byt< td=""><td>ок</td></byt<></addr></device></rw>	ок
es>[, <value>]</value>	Response
	+QIIC: <value></value>
	OK
Maximum Response Time	300ms

#### **Parameter**

<rw></rw>	0	Write command
	1	Read command
<device></device>	0-0xFF	Device address
<addr></addr>	0-0xFF	Register address
<bytes></bytes>	1-2	Read/write bytes
<value></value>	0-0xFFFF	Data value

**NOTE** 

Configuration will be not saved.



## 6.3. AT+CLVL Loud Speaker Volume Level

**AT+CLVL** is used to select the volume of the internal loudspeaker of the module.

AT+CLVL Loud Speaker Volume Level	
Test Command	Response
AT+CLVL=?	+CLVL: (list of supported <level>s)</level>
	OK
Read Command	Response
AT+CLVL?	+CLVL: <level></level>
	OK
Write Command	Response
AT+CLVL= <level></level>	OK
	If error is related to ME functionality:
	+CME ERROR: <err></err>
Maximum Response Time	300ms
Reference	
3GPP TS 27.007	

### **Parameter**

<level>

Integer type, value (0-100) with manufacturer specific range (Smallest value represents the lowest sound level), the default value of mode 0 is 80, the default value of mode 1 is 70 and the default value of mode 2 is 50.

#### **NOTES**

This parameter will not be saved.

#### 6.4. AT+QAUDGAIN Set Gain Offset

This command is to set gain offset to change uplink and downlink volume.

AT+QAUDGAIN Set Gain Offset	
Test Command	Response



AT+QAUDGAIN=?	+QAUDGAIN: (0-2),(-600~300),(-600~300)
	ок
Set Command	Response
AT+QAUDGAIN= <mode>[,<ul_gain_o< td=""><td>If configuration parameters are omitted:</td></ul_gain_o<></mode>	If configuration parameters are omitted:
ffset>, <dl_gain_offset>]</dl_gain_offset>	+QAUDGAIN: <mode>,<ul_gain_offset>,<dl_gain_offset></dl_gain_offset></ul_gain_offset></mode>
	OK
	If configuration parameters are entered:
	OK
	ERROR
Maximum Response Time	300ms

<mode></mode>	Numeric type, audio mode
	0 Handset
	1 Headset
	2 Handfree
<ul_gain_offset></ul_gain_offset>	Numeric type, value: -600~300, indicates uplink gain offset. Default value of mode
	0 is -1, the default value of mode 1 is 60, and the default value of mode 2 is 100.
<dl_gain_offset></dl_gain_offset>	Numeric type, value: -600~300, indicates downlink gain offset. Default value of
	mode 0 is 50, the default value of mode 1 is -1, and the default value of mode 2 is
	60.

**NOTE** 

This parameter will be saved to NV by default.

## 6.5. AT+QMIC Set Uplink(MIC) Volume

This command is to set MIC gains to change uplink volume.

AT+QMIC Set Uplink(MIC) Volume	
Test Command	Response
AT+QMIC=?	+QMIC: (0-2),(0-15)
	OK



Read Command AT+QMIC?	Response +QMIC: <handset volume="">,<headset volume="">,<handfreevolume> OK</handfreevolume></headset></handset>
Set Command AT+QMIC= <mode>,<volume></volume></mode>	Response OK ERROR
Maximum Response Time	300ms

<mode></mode>	Numeric typ	e, audio mode
---------------	-------------	---------------

0 Handset

1 Headset

Handfree

<volume>

Numeric type, indicates uplink volume, range: 0-15, the default value of mode 0 is 12, the default value of mode 1 is 14, and the default value of mode 2 is 13.

## NOTE

These parameters will not be saved.

## 6.6. AT+CMUT Mute Control

AT+CMUT is used to enable and disable the uplink voice mute during a voice call.

AT+CMUT Mute Control	
Test Command AT+CMUT=?	Response +CMUT: (0,1)  OK
Read Command AT+CMUT?	Response +CMUT: <n></n>
Write Command AT+CMUT= <n></n>	Response <b>OK</b>



	If error is related to ME functionality: +CME ERROR: <err></err>
Maximum Response Time	300ms
Reference 3GPP TS 27.007	

## NOTE

This parameter will not be saved.

## 6.7. AT+QCLKOUT Output Clock Source

AT+QCLKOUT is used to output clock source from pin25.

AT+QCLKOUT Output Clock Source	
Test Command AT+QCLKOUT=?	Response +QCLKOUT: (0,1),(1-4) OK
Read Command AT+QCLKOUT?	Response +QCLKOUT: <enable>,<source/>  OK +CME ERROR: <err></err></enable>
Write Command  AT+QCLKOUT= <enable>[,<source/>]</enable>	Response  OK  ERROR
Maximum Response Time	300ms



<enable></enable>	0	Disable to output clock source.
	1	Enable to output clock source.
<source/>	1	26MHz
	2	13MHz
	3	6.5MHz
	4	32.768KHz

### **NOTES**

- 1. Parameters will not be saved to NV.
- 2. When Pin25 is busy, it will return CME error.

## 6.8. AT+QSIDET Set the Side Tone Gain in the Current Mode

AT+QSIDET is used to set the side tone gain value of the current mode.

AT+QSIDET Set the Side Tone Gain in Current Mode		
Test Command AT+QSIDET=?	Response +QSIDET: (-450~200) OK	
Read Command AT+QSIDET?	Response +QSIDET: <stgain> OK</stgain>	
Write Command AT+QSIDET= <stgain></stgain>	Response OK ERROR	
Maximum Response Time	300ms	

#### **Parameter**

<stgain></stgain>	Numeric type, indicates the configured side tone gain in current mode. Range: -450~200.
	The default value of mode 0 is -100, the default value of mode 1 is -100, and the default
	value of mode 2 is -450.



Parameter will be saved to NV immediately by default.

## 6.9. AT+QAUDLOOP Enable Audio Loop Test

This command is used to enable audio loop test.

AT+QAUDLOOP Enable Audio Loop Test	
Test Command AT+QAUDLOOP=?	Response +QAUDLOOP: (0,1),(0-2) OK
Read Command AT+QAUDLOOP?	Response +QAUDLOOP: <enable>,<path>  OK</path></enable>
Set Command AT+QAUDLOOP= <enable>[,<path>]</path></enable>	Response OK ERROR
Maximum Response Time	300ms

#### **Parameter**

<enable> Numeric type; to enable or disable audio loop test

O Disable audio loop test

1 Enable audio loop test

<path> Compatible argument, no effect

**NOTE** 

These parameters will not be saved.



### 6.10. AT+QAUDMOD Set Audio Mode

AT+QAUDMOD sets the audio mode required for the connected device.

AT+QAUDMOD Set Audio Mode	
Test Command AT+QAUDMOD=?	Response +QAUDMOD: (0-2)
	ok
Read command	Response
AT+QAUDMOD?	+QAUDMOD: <mode></mode>
Write Command	Response
AT+QAUDMOD= <mode></mode>	OK  If error is related to ME functionality: +CME ERROR: <err></err>
Maximum Response Time	300ms

#### **Parameter**

<mode></mode>	Numeric type	e, indicates the	current configured	d audio mode

- 0 Echo canceller, noise suppressor, digital gain and calibration parameter for Handset
- 1 Echo canceller, noise suppressor, digital gain and calibration parameter for Headset
- 2 Echo canceller, noise suppressor, digital gain and calibration parameter for Speaker

#### **NOTE**

These parameters will not be saved.

## 6.11. AT+CRSL Set Ring Tone Volume

AT+CRSL can be used to set the volume of ring tone.

AT+CRSL Set Ring Tone Volume	
Test Command	Response
AT+CRSL=?	+CRSL: (list of supported <level>s)</level>



	ок
Read Command AT+CRSL?	Response +CRSL: <level></level>
Write Command AT+CRSL= <level></level>	Response OK ERROR
Maximum Response Time	300ms

#### **Parameter**

<level>

Numeric type, indicates the configured volume of ring tone. Range: 0-100, the default value of mode 0 is 80, the default value of mode 1 is 50, and the default value of mode 2 is 80.

# **NOTE**

This parameter will not be saved.

# 6.12. AT+CALM Mute the Ring Tone

AT+CALM is used to mute the ring tone. There is no ring tone when a call is coming.

AT+CALM Mute the Ring Tone	
Test Command	Response
AT+CALM=?	+CALM: (list of supported <n>s)</n>
	ОК
Read Command	Response
AT+CALM?	+CALM: <n></n>
	ОК
Write Command	Response
AT+CALM= <on></on>	ок
	ERROR
Maximum Response Time	300ms



#### **Parameter**

<n> Numeric type

- 0 Ring tone is normal
- 1 Ring tone is mute

# **NOTE**

This parameter will not be saved.

# 6.13. AT+QPCMON Enable PCM Sync & BCLK

AT+QPCMON is used to enable PCM sync & BCLK.

AT+QPCMON Enable PCM Sync	& BCLK
Test Command AT+QPCMON=?	Response +QPCMON: (list of supported <n>s)  OK</n>
Read Command AT+QPCMON?	Response +QPCMON: <n></n>
Write Command AT+QPCMON= <n></n>	Response OK ERROR
Maximum Response Time	300ms

# **Parameter**

<n> Numeric type

- O Disable PCM sync & BCLK
- 1 Enable PCM sync & BCLK

#### NOTE

This parameter will not be saved.



# 6.14. AT+QAUDCFG Audio Tuning Process

AT+QAUDCFG is used to query and configure various audio settings of UE.

AT+QAUDCFG Audio Tuning Pr	ocess
Test Command AT+QAUDCFG=?	Response  +QAUDCFG: "handset/eec", (list of supported <eecpara>s) +QAUDCFG: " headset/eec", (list of supported <eecpara>s) +QAUDCFG: "handfree/eec", (list of supported <eecpara>s) +QAUDCFG: "handset /nr/ul", (list of supported <nrpara>s) +QAUDCFG: "headset /nr/ul", (list of supported <nrpara>s) +QAUDCFG: "headset /nr/ul", (list of supported <nrpara>s) +QAUDCFG: "handfree/nr/ul", (list of supported <nrpara>s) +QAUDCFG: "handset/nr/dl", (list of supported <nrpara>s) +QAUDCFG: "headset /nr/dl ", (list of supported <nrpara>s) +QAUDCFG: "handfree/nr/dl ", (list of supported <nrpara>s) +QAUDCFG: "nau8814/dlgain", +QAUDCFG: "nau8814/ulgain", +QAUDCFG: "alc5616/dlgain", +QAUDCFG: "alc5616/ulgain", +QAUDCFG: "max9860/ulgain", +QAUDCFG: "max9860/ulgain",</nrpara></nrpara></nrpara></nrpara></nrpara></nrpara></nrpara></eecpara></eecpara></eecpara>
Maximum Response Time	300ms

# 6.14.1. AT+QAUDCFG="handset/eec" Set Echo Cancellation for Handset

AT+QAUDCFG="handset/eec" is used to set echo cancellation for handset.

AT+QAUDCFG="handset/eec"	Set Echo Ca	ncellation for Handset
Test Command		Response
AT+QAUDCFG=?		
		+QAUDCFG: "handset/eec", <eecpara></eecpara>



	ок
Write Command AT+QAUDCFG="handset/eec"[, <eecpara>]</eecpara>	Response  If configuration parameters are omitted: +QAUDCFG: "handset/eec", <eecpara></eecpara>
	ОК
	If configuration parameters are entered:
	OK ERROR

# **Parameter**

<eecpara></eecpara>	List of i	ntegers, separated by dot (.)
	Par0	Echocancollationcontrol word, 1: enable, 0: disable, default value might be different in different audio modes
	Par1	AEC (Acoustic Echo Canceller) control word, 1: enable, 0: disable, default value might be different in different audio modes
	Par2	EC_NrCoeffs_Real, Number of taps in real subband, range: 1-528. Default value might be different in different audio modes
	Par3	EC_NrCoeffs_Complex_1, Number of taps in complex subband 1, range: 1-528. Default value might be different in different audio modes
	Par4	EC_NrCoeffs_Complex_2, Number of taps in complex subband2, range: 1-528 Default value might be different in different audio modes
	Par5	EC_NrCoeffs_Complex_3, Number of taps in complex subband3, range: 1-528 Default value might be different in different audio modes
	Par6	EC_NrCoeffs_Complex_4, Number of taps in complex subband4, range: 1-528 Default value might be different in different audio modes
	Par7	EC_NrCoeffs_Complex_5, Number of taps in complex subband5, range: 1-528 Default value might be different in different audio modes
	Par8	EC_Blen, NLMS update block length range: (1,2,4,5,8). Default value might be different in different audio modes
	Par9	SER (Spectral Echo Reduction) control word, 1: enable, 0: disable, default value might be different in different audio modes
	Par10	SER_Beta, Smoothing factor of the echo reduction applied by the SER, range: 0-32767. Default value might be different in different audio modes
	Par11	SER_Overest, Overestimation factor of the echo estimation, range: 0-32767  Default value might be different in different audio modes
	Par12	SER_NoiseLevel, Overestimation factor of the noise estimation, range: 0-32767  Default value might be different in different audio modes
	Par13	SER_MultCtrl, Allowed amplitude of short-term variations of noise estimation range: 0-32767. Default value might be different in different audio modes



- **Par14** AGC (Automatic Gain Control) control word, 1: enable, 0: disable, default value might be different in different audio modes
- Par15 RE\_LdAddGain, Additionalgain, range: -16384~16384. Default value might be different in different audio modes
- Par16 RE\_LdGainLoLim, Total gain lower limit, range: 0-16384. Default value might be different in different audio modes
- **Par17** RE\_LdGainUpLim, Total gain upper limit, range: 0-16384. Default value might be different in different audio modes
- **Par18** RE\_NrShliFilt, Shifts for gain smoothing, range: -36~36. Default value might be different in different audio modes
- Par19 RE\_TriggerOffset, AGC trigger offset for the decision logic to allow more double-talk, range: -16384~16384. Default value might be different in different audio modes
- Par20 RE\_FarPowerCenter, Threshold on far-end power (below which overestimation is Par22, above which estimationisPar23), range: 0-16384. Default value might be different in different audio modes
- Par21 RE\_FarPowerDistorted, Threshold on far-end overestimated power (above which overestimated by Par24, to compensate residual echo when strong LS distortion), range: 0-16384. Default value might be different in different audio modes
- Par22 RE\_FarOverEstR0, Far-end power overestimation ratio if loudspeaker power is below Par20, range: 0-14666. Default value might be different in different audio modes
- Par23 RE\_FarOverEstR1, Far-end power overestimation ratio if loudspeaker power is abovePar20, range: 0-14666. Default value might be different in different audio modes
- Par24 RE\_FarOverEstR2, Far-end power overestimation ratio if loudspeaker power is above Par21, range: 0-16384. Default value might be different in different audio modes
- Par25 EDL (Echo Delay Lines) control word, 1: enable, 0: disable, default value might be different in different audio modes
- Par26 Delay in milliseconds in the echo reference path, range: 0-60. Default value might be different in different audio modes

#### **NOTES**

- 1. Auto save the configuration to NV. The setting is still valid after module restart.
- 2. If you set Par1 & Par9 & Par14 & Par25 to 0, please set Par0 to 0, otherwise it will return error. When you set Par0 to 0, Par1 & Par9 & Par14 & Par25 will be set to 0 by default.
- 3. Par2+2\*(Par3 Par4+Par5+Par6+Par7) < 2200.

#### **Example**

AT+QAUDCFG="handset/eec","1.1.100.100.100.1.1.1.2.1.32113.16384.2560.4096.1.0.0.5000.0.0.512
0.12288.256.282.307.1.0"

//Set configuration to default



OK
AT+QCFG="handset/eec" //Query
+QAUDCFG:
"handset/eec","1.1.100.100.100.1.1.1.2.1.32113.16384.2560.4096.1.0.0.5000.0.0.5120.12288.256.282.
307.1.0"
OK

#### 6.14.2. AT+QAUDCFG="headset/eec" Set Echo Cancellation for Headset

AT+QAUDCFG="headset/eec" is used to set echo cancellation for headset.

AT+QAUDCFG="headset/eec" Set Echo C	Cancellation for Headset
Test Command AT+QAUDCFG=?	Response +QAUDCFG: "headset/eec", <eecpara> OK</eecpara>
Write Command AT+QAUDCFG="headset/eec"[, <eecpara>]</eecpara>	Response If configuration parameters are omitted" +QAUDCFG: "headset/eec", <eecpara>  OK  If configuration parameters are entered"  OK  ERROR</eecpara>

#### **Parameter**

<eecpara> List of integers, separated by dot (.), for details, please refer to <eecpara> of
AT+QAUDCFG="handset/eec"

#### **Example**

AT+QAUDCFG="headset/eec","1.1.100.100.100.1.1.1.2.1.32113.16384.2560.4096.1.0.0.5000.0.0.512
0.12288.256.282.307.1.0"

//Set configuration to default

# OK

AT+QCFG="headset/eec" //Query



#### +QAUDCFG:

 $"headset/eec", "1.1.100.100.100.1.1.1.2.1.32113.16384.2560.4096.1.0.0.5000.0.0.5120.12288.256.282.\\ 307.1.0"$ 

OK

# 6.14.3. AT+QAUDCFG="handfree/eec" Set Echo Cancellation for Handfree

AT+QAUDCFG="handfree/eec" is used to set echo cancellation for handfree.

FG: "handfree/eec", <eecpara></eecpara>
ation parameters are omitted:  FG="handfree/eec", <eecpara>  ation parameters are entered:</eecpara>

#### **Parameter**

**<eecpara>** List of integers, separated by dot (.), for more details, please refer to **<eecpara>** of AT+QAUDCFG="handset/eec".

#### **Example**

AT+QAUDCFG="handfree/eec","1.1.150.150.150.1.1.1.2.1.32113.16384.10240.16384.1.1800.80.8192
.0.0.6144.12288.256.282.307.1.0"
//Set configuration to default

#### OK

AT+QCFG="handfree/eec"

//Query

#### +QAUDCFG:

"handfree/eec","1.1.150.150.150.1.1.1.2.1.32113.16384.10240.16384.1.1800.80.8192.0.0.6144.12288. 256.282.307.1.0"

OK



# 6.14.4. AT+QAUDCFG="handset/nr/ul" Set Uplink Noise Reduction for Handset

AT+QAUDCFG="handset/nr/ul" is used to set uplink noise reduction for handset.

AT+QCFG="handset/nr/ul" Set Uplink Nois	se Reduction for Handset
Test Command AT+QAUDCFG=?	Response +QAUDCFG: "handset/nr/ul", <nrpara> OK</nrpara>
Write Command AT+QAUDCFG="handset/nr/ul"[, <nrpara>]</nrpara>	Response If configuration parameters are omitted: +QAUDCFG: "handset/nr/ul", <nrpara>  OK  If configuration parameters are entered: OK ERROR</nrpara>

# **Parameter**

<nrpara></nrpara>	List of	integers, separated by dot (.).
	Par0	Noise Reduction control word, 1: enable, 0: disable, Default value might be different in different audio modes
	Par1	NR_AttenFactorMinVal, Minimum value of Noise Reduction gain, range: 0-32767. Default value might be different in different audio modes
	Par2	NR_OvEstFacBandZero, Overestimation factor for Noise Reduction in band 0, range: 0-32767. Default value might be different in different audio modes
	Par3	NR_OvEstFacBandNoZero, Overestimation factor for Noise Reduction in band 1 to 7, range: 0-32767. Default value might be different in different audio modes
	Par4	NR_GainFactor, Defines how aggressive the NR is, range: 0-32767. Default value might be different in different audio modes
	Par5	NR_GainLimit, Maximum gain allow, range: 0-32767. Default value might be different in different audio modes

#### NOTE



AT+QAUDCFG="handset/nr/ul", "1.6000.9000.10000.1800.19660" //Set configuration to default

OK
AT+QAUDCFG="handset/nr/ul" //Query

+QAUDCFG: "handset/nr/ul", "1.6000.9000.10000.1800.19660"

OK

#### 6.14.5. AT+QAUDCFG="handset/nr/dl" Set Down Link Noise Reduction for Handset

AT+QAUDCFG="handset/nr/dl" is used to set down link noise reduction for handset.

AT+QCFG="handset/nr/dl" Set Down Link	Noise Reduction for Handset
Test Command AT+QAUDCFG=?	Response +QAUDCFG: "handset/nr/dl", <nrpara> OK</nrpara>
Write Command AT+QAUDCFG="handset/nr/dl"[, <nrpara>]</nrpara>	Response If configuration parameters are omitted: +QAUDCFG:"handset/nr/dl", <nrpara>  OK  If configuration parameters are entered: OK ERROR</nrpara>

#### **Parameter**

<nrpara></nrpara>	List of integers, separated by dot (.).	For more details, please refer to <nrpara> of AT+</nrpara>
	QAUDCFG="handset/nr/ul"	

#### **NOTE**



#### 6.14.6. AT+QAUDCFG="headset/nr/dl" Set Down Link Noise Reduction for Headset

AT+QAUDCFG="headset/nr/dl" is used to set down link noise reduction for headset.

AT+QCFG="headset/nr/dl" Set Down Link Noise Reduction for Headset	
Test Command AT+QAUDCFG=?	Response +QAUDCFG: "headset/nr/dl", <nrpara> OK</nrpara>
Write Command AT+QAUDCFG="headset/nr/dl"[, <nrpara>]</nrpara>	Response If configuration parameters are omitted: +QAUDCFG: "headset/nr/dl", <nrpara>  OK  If configuration parameters are entered: OK ERROR</nrpara>

#### Parameter

<nrpara> List of integers, separated by dot (.). For details, please refer to <nrpara> of AT+ QAUDCFG="handset/nr/ul"

#### **NOTE**



AT+QAUDCFG="headset/nr/dl","1.12000.10000.10000.19660" //Set configuration to default

OK
AT+QAUDCFG="headset/nr/dl" //Query

+QAUDCFG: "headset/nr/dl", "1.12000.10000.10000.19660"

OK

# 6.14.7. AT+QAUDCFG="headset/nr/ul" Set Uplink Noise Reduction for Headset

AT+QAUDCFG="headset/nr/ul" is used to set uplink noise reduction for headset.

AT+QCFG="headset/nr/ul" Set Uplink Noise Reduction for Headset	
Test Command AT+QAUDCFG=?	Response +QAUDCFG: "headset/nr/ul", <nrpara> OK</nrpara>
Write Command AT+QAUDCFG="headset/nr/ul"[, <nrpara>]</nrpara>	Response If configuration parameters are omitted: +QAUDCFG: "headset/nr/ul", <nrpara>  OK  If configuration parameters are entered: OK ERROR</nrpara>

#### Parameter

<nrpara> List of integers, separated by dot (.). For details, please refer to <nrpara> of AT+ QAUDCFG="handset/nr/ul"

# **NOTE**



# 6.14.8. AT+QAUDCFG="handfree/nr/dl" Set Down Link Noise Reduction for Handfree

AT+QAUDCFG="handfree/nr/dl" is used to set down link noise reduction for handfree.

AT+QCFG="handfree/nr/dl" Set Down Lin	k Noise Reduction for Handfree
Test Command AT+QAUDCFG=?	Response +QAUDCFG: "handfree/nr/dl", <nrpara> OK</nrpara>
Write Command  AT+QAUDCFG="handfree/nr/dl"[, <nrpara>]</nrpara>	Response If configuration parameters are omitted: +QAUDCFG: "handfree/nr/dl", <nrpara>  OK</nrpara>
	If configuration parameters are entered:  OK  ERROR

#### **Parameter**

#### **NOTE**



AT+QAUDCFG="handfree/nr/dl","1.12000.10000.10000.19660" //Set configuration to default

OK
AT+QAUDCFG="handfree/nr/dl" //Query

+QAUDCFG: "handfree/nr/dl","1.12000.10000.10000.19660"

OK

# 6.14.9. AT+QAUDCFG="handfree/nr/ul" Set Uplink Noise Reduction for Handfree

AT+QAUDCFG="handfree/nr/ul" is used to set uplink reduction for handfree.

AT+QCFG="handfree/nr/ul" Set Uplink Noise Reduction for Handfree	
Test Command AT+QAUDCFG=?	Response +QAUDCFG: "handfree/nr/ul", <nrpara> OK</nrpara>
Write Command AT+QAUDCFG="handfree/nr/ul"[, <nrpara>]</nrpara>	Response If configuration parameters are omitted: +QAUDCFG: "handfree/nr/ul", <nrpara>  OK  If configuration parameters are entered: OK ERROR</nrpara>

#### **Parameter**

#### **NOTE**



AT+QAUDCFG="handfree/nr/ul","1.6000.9000.10000.1800.19660" //Set configuration to default

OK
AT+QAUDCFG="handfree/nr/ul" //Query

+QAUDCFG: "handfree/nr/ul","1.6000.9000.10000.1800.19660"

OK

# 6.14.10. AT+QAUDCFG="nau8814/dlgain" Set Down Link Gain Level for Codec nau8814

AT+QAUDCFG="nau8814/dlgain" is used to set down link gain level for codec nau8814.

AT+QAUDCFG="nau8814/dlgain" Set Dow	n Link Gain Level for Codec nau8814
Test Command AT+QAUDCFG=?	Response +QAUDCFG: "nau8814/dlgain", <level> OK</level>
Write Command AT+QAUDCFG="nau8814/dlgain"[, <level>]</level>	Response If configuration parameters are omitted: (+QAUDCFG="nau8814/dlgain"),  Return current configuration: +QAUDCFG: "nau8814/dlgain", <level> OK  If configuration parameters are entered: OK ERROR  If error is related to ME functionality: +CME ERROR: <err></err></level>

#### **Parameter**

**<Level>** Numeric type, indicates the downlink gain of nau8814, range: 0-100, default: 100



This parameter will not be saved.

# **Example**

AT+QAUDCFG="nau8814/dlgain",85 //Set downlink gain to 85

OK

# 6.14.11. AT+QAUDCFG="nau8814/ulgain" Set Uplink Gain Level for Codec nau8814

AT+QAUDCFG="nau8814/ulgain" is used to set uplink gain level for codec nau8814.

AT+QAUDCFG="nau8814/ulgain" Set Up	olink Gain Level for Codec nau8814
Test Command AT+QAUDCFG=?	Response
	+QAUDCFG: "nau8814/ulgain", <level></level>
	ок
Write Command	Response
AT+QAUDCFG="nau8814/ulgain"[, <level>]</level>	If configuration parameters are omitted:
	(+QAUDCFG="nau8814/ulgain"),
	Return current configuration:
	+QAUDCFG: "nau8814/ulgain", <level></level>
	If configuration parameters are entered:
	ОК
	ERROR
	If error is related to ME functionality:
	+CME ERROR: <err></err>

#### **Parameter**

**<Level>** Numeric type, indicates the uplink gain of nau8814, range: 0-100, default: 84



This parameter will not be saved.

# **Example**

AT+QAUDCFG="nau8814/ulgain",90 //Set uplink gain to 90

OK

# 6.14.12. AT+QAUDCFG="alc5616/dlgain" Set Down Link Gain Level for Codec alc5616

AT+QAUDCFG="alc5616/dlgain" is used to set down link gain level for codec alc5616.

AT+QAUDCFG="alc5616/dlgain" Set Dow	n Link Gain Level for Codec alc5616
Test Command AT+QAUDCFG=?	Response +QAUDCFG: "alc5616/ dlgain", <level> OK</level>
Write Command AT+QAUDCFG="alc5616/dlgain"[, <level>]</level>	Response If configuration parameters are omitted: (+QAUDCFG="alc5616/dlgain"),  Return current configuration: +QAUDCFG: "alc5616/dlgain", <level> OK  If configuration parameters are entered: OK ERROR</level>
	If error is related to ME functionality: +CME ERROR: <err></err>

# **Parameter**

**Level>** Numeric type, indicates the downlink gain of alc5616, range: 0-100, default: 100



This parameter will not be saved.

# **Example**

AT+QAUDCFG="alc5616/dlgain",85

//Set downlink gain to 85

OK

# 6.14.13. AT+QAUDCFG="alc5616/ulgain" Set Uplink Gain Level for Codec alc5616

AT+QAUDCFG="alc5616/ulgain" is used to set uplink gain level for codec alc5616.

AT+QAUDCFG="alc5616/ulgain" Set Upl	ink Gain Level for Codec alc5616
Test Command AT+QAUDCFG=?	Response
	+QAUDCFG: "alc5616/ ulgain", <level></level>
	ок
Write Command	Response
AT+QAUDCFG="alc5616/ulgain"[, <level>]</level>	If configuration parameters are omitted:
	(+QAUDCFG="alc5616/ulgain"),
	Return current configuration:
	+QAUDCFG: "alc5616/ulgain", <level></level>
	If configuration parameters are entered:
	ОК
	ERROR
	If error is related to ME functionality:
	+CME ERROR: <err></err>

# **Parameter**

**Level>** Numeric type, indicates the uplink gain of alc5616, range: 0-100, default: 74



This parameter will not be saved.

# **Example**

AT+QAUDCFG="alc5616/ulgain",85 //Set uplink gain to 85

OK

# 6.14.14. AT+QAUDCFG="max9860/dlgain" Set Down Link Gain Level for Codec max9860

AT+QAUDCFG="max9860/dlgain" is used to set down link gain level for codec max9860.

AT+QAUDCFG="max9860/dlgain" Set Down Link Gain Level for Codec max9860	
Test Command AT+QAUDCFG=?	Response +QAUDCFG: "max9860/dlgain", <level> OK</level>
Write Command AT+QAUDCFG="max9860/dlgain"[, <level>]</level>	Response If configuration parameters are omitted: (+QAUDCFG="max9860/dlgain"),  Return current configuration: +QAUDCFG: "max9860/dlgain", <level> OK  If configuration parameters are entered: OK ERROR</level>
	If error is related to ME functionality: +CME ERROR: <err></err>

# **Parameter**

**Level>** Numeric type, indicates the down link gain of max9860, range: 0-100, default: 87



This parameter will not be saved.

# **Example**

AT+QAUDCFG="max9860/dlgain",85 //Set downlink gain to 85

OK

# 6.14.15. AT+QAUDCFG="max9860/ulgain" Set Uplink Gain Level for Codec max9860

AT+QAUDCFG="max9860/ulgain" is used to set uplink gain level for codec max9860.

AT+QAUDCFG="max9860/ulgain" Set Uplink Gain Level for Codec max9860	
Test Command AT+QAUDCFG=?	Response +QAUDCFG: "max9860/ulgain", <level> OK</level>
Write Command AT+QAUDCFG="max9860/ulgain"[, <level>]</level>	Response If configuration parameters are omitted: (+QAUDCFG="max9860/ulgain"),  Return current configuration: +QAUDCFG: "max9860/ulgain", <level> OK  If configuration parameters are entered: OK ERROR  If error is related to ME functionality:</level>
	+CME ERROR: <err></err>

# **Parameter**

**Level>** Numeric type, indicates the uplink gain of max9860, range: 0-100, default: 80



This parameter will not be saved.

# **Example**

AT+QAUDCFG="max9860/ulgain",85 //Set uplink gain to 85

OK

**AUDCFG="handset/nr/dl","1.12000.10000.10000.1000.19660"** //Set configuration to default

OK

AT+QAUDCFG="handset/nr/dl" //Query

+QAUDCFG: "handset/nr/dl","1.12000.10000.10000.1000.19660"

OK



# **7** Appendix

#### **Table 3: Related Documents**

SN	Document name	Remark
[1]	Quectel_UGxx_Hardware_Design	UG95 and UG96 Hardware Design
[2]	Quectel_UGxx_Reference_Design	UG95 and UG96 Reference Design
[3]	Quectel_WCDMA_UGxx_AT_Commands_ Manual	UGxx AT Commands Manual

# **Table 4: Terms and Abbreviations**

Abbreviation	Description
LDO	Low Dropout Regulator
ME	Mobile Equipment (UG95 and UG96 Module)
NV	Non-Volatile Memory
PCM	Pulse-code Modulation