

SIMCOM WCDMA Wireless Module

SIM5xxx Audio Application Note



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Version history

Date	Version	Description of change	Author
2010-03-23	1.01	Origin	Libing
2011-08-26	1.02	Add SIM5320 and ECHO description	Libing



1 Introduction

SIM52XX provides some AT commands for audio tuning. This document describes how to design and tune the audio part for best performance of SIMCOM WCDMA wireless module. (SIM52XX represents the series which is stated below.)

2 Scope of the document

This document is intended for the following versions of the SIMCom modules

- •SIM5215/SIM5216
- •SIM5218
- •SIM5320



3 Audio channel overview

The table below shows the audio channels of different SIMCOM WCDMA wireless modules.

1110 0001001			
Module	Au	ıdio Channel	Note
SIM5215	Handset:	Input: MIC1_P, MIC1_N	
SIM5216 SIM5218	AT+CSDVC =1	Output: EAR_P, EAR_N	
511113216	Headset:	Input: HP_MIC	
	AT+CSDVC=2	Output: HPR, HPL	
	Handfree:	Input: MIC1_P, MIC1_N	
	AT+CSDVC=3	Output: SPK_P,SPK_N	
	PCM: AT+CSDVC=4	PCM_SYNC PCM_CLK PCM_DOUT PCM_DIN	
SIM5320	Handset:	Input: MIC1_P, MIC1_N	
	AT+CSDVC =1	Output: EAR_P, EAR_N	
	Handfree:	Input: MIC1_P, MIC1_N	
	AT+CSDVC=3	Output: SPK_P,SPK_N	
	PCM:	PCM_SYNC	
	AT+CSDVC=4	PCM_CLK	
		PCM_DOUT	
		PCM_DIN	



4 Hardware Design

4.1 Speaker interface configuration

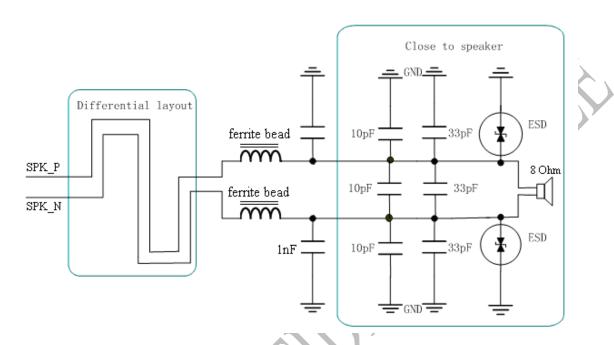


Figure 1: Speaker interface configuration

Because SPK_P and SPK_N are outputs of Class-D audio amplifier, optional EMI filtering is shown at Figure 1; these components (two ferrite beads and two capacitors) can be added to reduce electromagnetic interference. If used, they should be located near the SPK_P and SPK_N. Considerable current flows between the audio output pins and the speaker, so wide PCB traces are recommended (~ 20 mils). 80hm speaker is suggested. And the SPK_P and SPK_N should layout differential, and they should be far away from VBAT, RF signals, clock and other high power or high frequency signals.

4.2 Receiver interface configuration



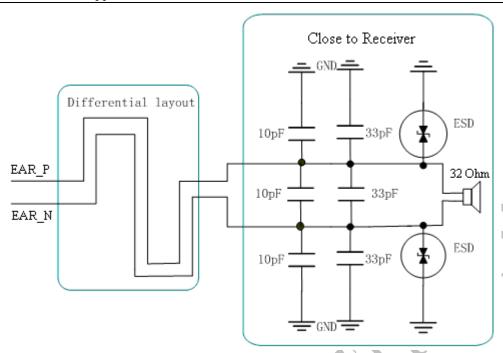


Figure 2: Receiver interface configuration

33p and 10p are suggested to be added beside the 32 Ohm receiver to reduce RF interfere. The width of EAR_P and EAR_N lines is typical 6 mils to reduce impedance. They should be far away from VBAT, RF signals, clock and other high power or high frequency signals. EAR_P and EAR_N lines should be layout differential.

4.3 Microphone interfaces configuration

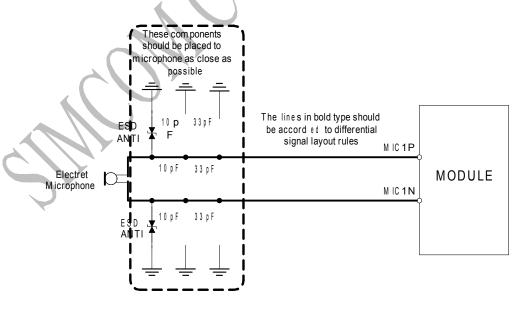


Figure 3: Microphone interface configuration



SIMCOM WCDMA wireless modules have integrated internal MIC bais circuit. There is no need to pull the MIC1_P and MIC_M up to the external power, because they have been pulled up in the Module. MIC_P and MIC_N should be layout differential.

4.4 External MIC bias circuit configuration

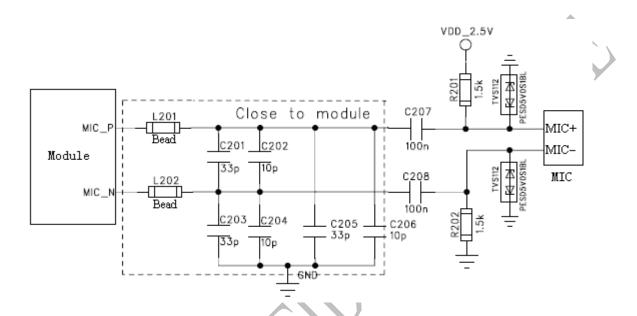


Figure 4: External MIC bias circuit configuration

External MIC bias circuit is described in the figure above. This figure can be compatible with all SIMCOM WCDMA wireless modules. MIC_P and MIC_N should be layout differential.



4.5 Earphone interface configuration

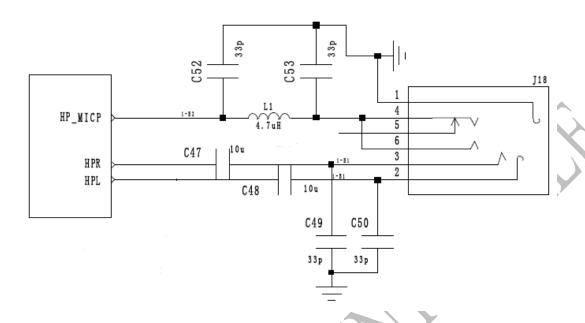


Figure 5: Earphone interface configuration

HPR and HPL are not difference signals. So they should routine alone and keep separate from each other. And C47 and C48 are required to isolate the DC signal.

4.6 Referenced electronic characteristic

Table 10: MIC Input Characteristics

Parameter	Min	Тур	Max	Unit
Working Voltage		1.80		V
Working Current	70		400	uA
External Microphone Load Resistance	1.2	2.2		k Ohms

Table 11: Audio Output Characteristics

Parameter		Min	Тур	Max	Unit	
Normal Output (EAR P,EAR Differential	load Resistance	27	32		Ohm	
N)	Billerential	Output power		70		mW
Auxiliary Output	Single Ended	load Resistance	12	16		Ohm

(HPR,HPL)

Dif	tterential	oad Resistance	27	32	Ohm
Sin		Output oower		21.6	mW

Table 12: Speaker Output Characteristics

Parameter	Min	Тур	Max	Unit
Quiescent Current		6.2		mA
Output power(1KHz)		500		mW

5 Audio Tuning

The audio programming model shows how the signal path can be influenced by varying AT command parameters. Parameters <micAmp>, <txGain> , <txVol>, <txFilter>, <rxGain> , <stGain> , <rxVol> and <rxFilter> can be adjusted with corresponding AT commands. For more information on the AT commands and parameters see Section 3.8.5.1.

NOTE: Please reference document [1] for detailed information of each AT command.

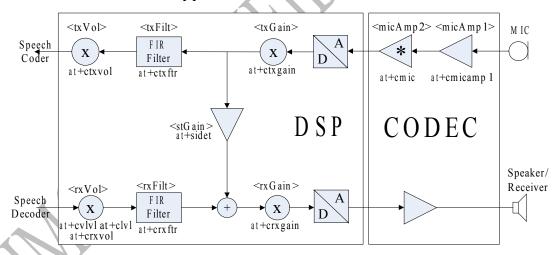


Figure 6: Audio programming model

- For SIM5320/SIM5215/SIM5216, there is no <micAmp2>, so at+cmic does not function.
- The parameters located in DSP, will not function until the next calling setup, such as at+ctxvol, at+ctxftr, at+ctxgain, at+clvl, at+cvlvl, at+crxftr, at+crxgain, at+sidet. And the parameters located in codec can function in calling.



5.1 MIC volume and frequency response

In figure2, one can turn adjust codec part or DSP part parameters to get desired MIC volume or frequency response.

Codec part

<micamp1>: AT+cmicamp1 (Detail description refer to 5.1.1) <micamp2>: AT+cmic (Detail description refer to 5.1.2)

DSP part

<txGain>: AT+ctxgain (Detail description refer to 5.1.3)
<txFilt>: AT+ctxftr (Detail description refer to 5.1.4)
<txVol>: AT+ctxvol (Detail description refer to 5.1.5)

Note: From figure2, one can see that AT+cmicamp1, AT+cmic, AT+ctxgain, AT+ctxftr, AT+ctxvol can influence sidetone.

5.1.1 AT+CMICAMP1 Set value of micamp1

Description

The command is used to set audio path parameter – micamp1; this is different with AT+CMIC. With this command you can change the first stage of MIC amplify value based on your design separately and refer to related hardware design document to get more information

SIM PIN	References
NO	Vendor

Syntax

	Test Command	Responses
	AT+CMICAMP1=?	+CMICAMP1: (list of supported <amp_val>s)</amp_val>
		OK
	Read Command	Responses
	AT+ CMICAMP1?	+CMICAMP1: <amp_val></amp_val>
		OK
d	Write Command	Responses
	AT+CMICAMP1= <amp_val></amp_val>	OK
١		ERROR

Defined values

<amp_val>
amplify value number which is from 0 to 1. 0 is 0DB and 1 is 24DB.

Examples



```
AT+CMICAMP1=0

+CMICAMP1: 0

OK

AT+CMICAMP1: 0

OK

AT+ CMICAMP1=?

+CMICAMP1: (0-1)

OK
```

5.1.2 AT+CMIC Microphone volume control

Description

The command is used to control the microphone gain level. When the Module restarts, the gain level will resume as default values. The setting will be saved to nonvolatile memory after write command is executed. There is at+cmic in /SIM5215/SIM5216/SIM5320.

SIM PIN	References
NO	Vendor

Syntax

Test Command	Responses
AT+CMIC=?	+CMIC: (list of supported <gainlevel>s)</gainlevel>
	OK
Read Command	Responses
AT+CMIC?	+CMIC: <gainlevel></gainlevel>
	OK
Write Command	Responses
AT+CMIC= <gainlevel></gainlevel>	OK
	ERROR

Defined values

<gainlevel>

Range from 0 to 15, and 0 is the lowest gain level.

When the audio output of device is handset, 7 is default value; when headset, 7 is default value; when speaker, 4 is default value.

Examples

```
AT+CMIC=5
OK
```



```
AT+CMIC?
+CMIC:5
OK
```

5.1.3 AT+CTXGAIN Set TX gain

Description

The command is used to set audio path parameter – TX gain, and refer to related hardware design document to get more information.

SIM PIN	References
NO	Vendor

Syntax

Test Command	Responses
AT+CTXGAIN=?	+CTXGAIN: (list of supported <tx_gain>s)</tx_gain>
	OK
Read Command	Responses
AT+CTXGAIN?	+CTXGAIN: <tx_gain></tx_gain>
	OK
Write Command	Responses
AT+CTXGAIN= <tx_gain></tx_gain>	OK

Defined values

```
<tx_gain>
TX gain level which is from 0 to 65535.
```

Examples

```
AT+CTXGAIN=1234
OK
```

5.1.4 AT+CTXFTR Set TX filter

Description

The command is used to set audio path parameter – TX filter, and refer to related hardware design document to get more information.

SIM PIN	References
NO	Vendor

Syntax

Test Command	Responses
AT+CTXFTR=?	+CTXFTR: (list of supported <tx_ftr_n>s)</tx_ftr_n>
	OK
Read Command	Responses
AT+CTXFTR?	+CTXFTR: <tx_ftr_></tx_ftr_> ,<>, <tx_ftr_></tx_ftr_>
	OK
Write Command	Responses
AT+CTXFTR=	OK
<tx_ftr_></tx_ftr_> ,<>, <tx_ftr_></tx_ftr_>	

Defined values

```
<tx_ftr_X>
TX filter level which is from 0 to 65535. (N is from 0 to 7)
```

Examples

```
AT+CTXFTR=1111,2222,3333,4444,5555,6666,7777

OK
```

5.1.5 AT+CTXVOL Set TX volume

Description

The command is used to set audio path parameter – TX volume, and refer to related hardware design document to get more information.

SIM PIN	References	
NO	Vendor	

Syntax

Test Command	Responses
AT+CTXVOL=?	+CTXVOL: (list of supported <tx_vol>s)</tx_vol>
	OK
Read Command	Responses
AT+CTXVOL?	+CTXVOL: <tx_vol></tx_vol>
	OK
Write Command	Responses
AT+CTXVOL= <tx_vol></tx_vol>	OK

Defined values



< tx vol >

TX volume level which is from 0 to 65535.

Examples

```
AT+CTXVOL=1234
OK
```

5.2 Receiver or Speaker volume and frequency response

In figure 2, one can only turn adjust DSP part parameters to get desired receiver or speaker volume and frequency response. The parameter of codec part of module can not be adjusted.

DSP part

AT+crvvol (Detail description refer to)

Note: From figure2, one can see that at+crxgain can influence sidetone.

AT+crxvol is used for fine tuning for <rxVol>. AT+CLVL and AT+CVLVL are used for coarse tuning for <rxVol>. Although we provide some AT for adjust the volume such as CRXVOL and CRXGAIN. These commands can change the voice levels together, that is to say, all the levels are promoted by these two parameters. But if you want to change each sound level value, you should use command CVLVL.

AT+CVLVL This command changes the sound level values of the command CLVL. Now we provide 5 levels for each audio channel. The level 0 is muted and it can not be changed by CVLVL. Levels 1 to 4 are supported to change the value of sound level. CVLVL command could let you change these four levels. The bigger the number presents the louder the voice. And the range of each level is -5000 to 5000.

5.2.1 AT+CRXGAIN Set RX gain

Description

The command is used to set audio path parameter – RX gain, and refer to related hardware design document to get more information.

SIM PIN	References
NO	Vendor

Syntax



AT+CRXGAIN=?	+CRXGAIN: (list of supported <rx_gain>s) OK</rx_gain>
Read Command	Responses
AT+CRXGAIN?	+CRXGAIN: <rx_gain> OK</rx_gain>
Write Command	Responses
AT+CRXGAIN= <rx_gain></rx_gain>	OK

Defined values

```
<rx_gain>
RX gain level which is from 0 to 65535.
```

Examples

```
AT+CRXGAIN=1234
OK
```

5.2.2 AT+CRXFTR Set RX filter

Description

The command is used to set audio path parameter – RX filter, and refer to related hardware design document to get more information.

SIM PIN	References
NO	Vendor

Syntax

Test Command	Responses
AT+CRXFTR=?	+CRXFTR: (list of supported <rx_ftr_n>s)</rx_ftr_n>
	OK
Read Command	Responses
AT+CRXFTR?	+CRXFTR: <rx_ftr_></rx_ftr_> ,<>, <rx_ftr_></rx_ftr_>
	OK
Write Command	Responses
AT+CRXFTR=	OK
<rr_ftr_></rr_ftr_> ,<>, <rr_ftr_></rr_ftr_>	

Defined values

<rx_ftr_<i>X></rx_ftr_<i>	
RX filter level which is from 0 to 65535. (N is from 0 to 7)	



Examples

AT+CRXFTR=1111,2222,3333,4444,5555,6666,7777

5.2.3 AT+CVLVL Set value of sound level

Description

The command is used to set audio path parameter – RX volume; this command is different from CRXVOL, command CRXVOL will modify the values of all sound levels offset we provided together. With this command you can change the value of each sound level based on your design separately and refer to related hardware design document to get more information.

AT+crvvol is used for fine tuning for <rxVol>. AT+CLVL and AT+CVLVL are used for coarse tuning for <rxVol>. Although we provide some AT for adjust the volume such as CRXVOL and CRXGAIN. These commands can change the voice levels together, that is to say, all the levels are promoted by these two parameters. But if you want to change each sound level value, you should use command CVLVL.

SIM PIN	References
NO	Vendor

Syntax

Test Command	Responses
AT+CVLVL=?	+CVLVL: (list of supported <lvl>s),(list of supported <lvl_v-alue>s) OK</lvl_v-alue></lvl>
Read Command	Responses
AT+CVLVL?	+CVLVL: <lvl_value1>,<lvl_value2>,<lvl_value3>,<lvl_value4></lvl_value4></lvl_value3></lvl_value2></lvl_value1>
	OK
Write Command	Responses
AT+CVLVL= < lvl>,	OK
<lul>value></lul>	ERROR

Defined values

```
sound level number which is from 1 to 4.

<lvl_value>
    sound level value which is from -5000 to 5000.

<lvl_value1>
    sound level value that sound level number equals 1.

<lvl_value2>
    sound level value that sound level number equals 2.

<lvl_value3>
    sound level value that sound level number equals 3.

<lvl_value4>
```



sound level value that sound level number equals 4.

Examples

```
AT+CVLVL=1,-2000

+CVLVL:-2000

OK

AT+CVLVL?

+CVLVL:-2000,-200,500,1000

OK

AT+CVLVL=?

+CVLVL: (1-4),(-5000~5000)

OK
```

5.2.4 AT+CLVL Loudspeaker volume level

Description

Write command is used to select the volume of the internal loudspeaker audio output of the device.

Test command returns supported values as compound value.

SIM PIN	References
NO	3GPP TS 27.007

Syntax

Test Command	Responses
AT+CLVL=?	+CLVL: (list of supported <level>s) OK</level>
Read Command	Responses
AT+CLVL?	+CLVL: <level> OK</level>
Write Command	Responses
AT+CLVL= <level></level>	OK
	ERROR

Defined values

<level>

Integer type value which represents loudspeaker volume level. The range is from 0 to 4, and 0 represents the lowest loudspeaker volume level, 2 is default factory value.

NOTE < level> is nonvolatile, and it is stored when restart.

Examples



```
AT+CLVL?
+CLVL:2
OK
AT+CLVL=3
OK
```

5.2.5 AT+CRXVOL Set RX volume

Description

The command is used to set audio path parameter – RX volume, and refer to related hardware design document to get more information.

SIM PIN	References
NO	Vendor

Syntax

Test Command	Responses
AT+CRXVOL=?	+CRXVOL: (list of supported <rx_vol>s)</rx_vol>
	OK
Read Command	Responses
AT+CRXVOL?	+CRXVOL: <rx_vol></rx_vol>
	OK
Write Command	Responses
AT+CRXVOL= <rx_vol></rx_vol>	OK

Defined values

```
<rx_vol>
RX volume level which is from -100 to 100.
```

Examples

```
AT+CRXVOL=12
OK
```

5.3 AT+SIDET Digital attenuation of sidetone

Description

The command is used to set digital attenuation of sidetone. For more detailed information, please refer to relevant HD document.



NO Vendor

Syntax

Test Command	Responses
AT+SIDET=?	+SIDET: (list of supported <st>s)</st>
	OK
Read Command	Responses
AT+SIDET?	+SIDET: <st></st>
	OK
Write Command	Responses
AT+SIDET= <st></st>	OK
	ERROR

Defined values

<st>

Digital attenuation of sidetone, integer type in decimal format and nonvolatile.

Range: from 0 to 65535.

Factory value: HANDSET:2034, HEADSET:1024, SPEAKER PHONE: 0.

Examples

```
AT+CSDVC=1
OK
AT+SIDET?
+SIDET: 2304
OK
```

5.4 Echo canceller

SIM52xx has AT command "AT+ CLEDITST" to adjust echo canceller.

This AT command is used to select the echo cancellation mode. Each audio channel has it's own default echo cancellation mode.

at+cecm=0 : disable EC mode

at+cecm=1 : EC mode recommended for Handset

at+cecm=2: EC mode recommended for Headset

at+cecm=3: EC mode recommended for HANDSFREE

at+cecm=4: EC mode recommended for SPEAKER

at+cecm=5: EC mode recommended for BT HEADSET

at+cecm=6, EC mode recommended for Speaker phone aggressive echo arithmetic

at+cecm=7, EC mode recommended for Speaker phone medium echo arithmetic

at+cecm=8, EC mode recommended for Speaker least aggressive echo arithmetic



5.5 TDD noise

Making sure the module connect to ground well can help to reduce the TDD noise and improve ESD.

Below we take SIM5218 as an example to explain how to reduce TDD niose. There are 4 out-feet feet of SIM5218A bottom shield. One can solder them to bare cooper on host board, so that SIM5218A can be connected to the ground well. Figure 7 is SIM5218 PCB decal we suggest. There a large piece of cooper to help connecting well with SIM5218 bottom shield.

Because the height of 70- pins plug on SIM5218A is 1.5mm, in order to better connecting the SIM5218A with to the ground of the host board well, it is suggested to choose 2.0mm height 70-pins socket for host board. NAS AXK770247G is suitable. 2.5mm will be too high, it will be suspending above host board. 1.5mm will be too tight.

Filtering capacitors and beads are suggested to be added in the audio lines, 33p and 10p can help reduce the 850Mhz and 1800Mhz RF interfere. If it is signal, the filtering capacitors and beads are suggested to add beside the module pins. If it is output trace, the filtering capacitors and beads are suggested to add beside the handset/headset/speaker connector.

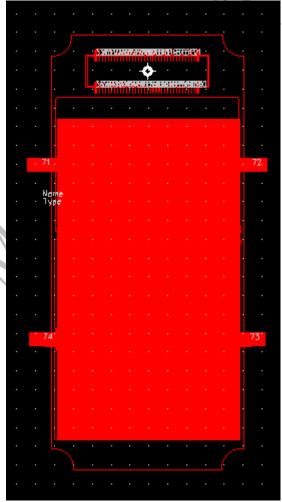


Figure 7: SIM5218 PCB decal

When one mounts SIM5218A on host board, please pay attention to that the pin sequence of SIM5218A is mirror image of pin sequence of 70-pin socket connector on host.



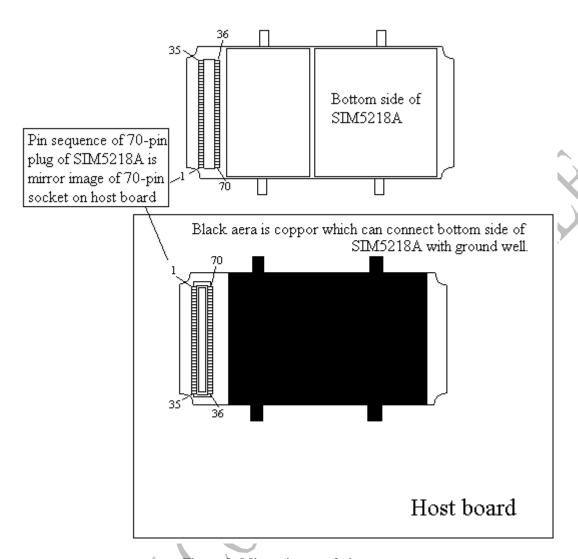


Figure 8: Mirror image of pin sequence

5.6 Sending and receiving distortion

There are many factors which may influence the sending and receiving distortion.

- 1. Unsuitable FIR parameters. They can be adjust by AT+CTXGAIN, AT+CRXGAIN.
- 2. Too large TDD noise. Please refer to 5.5 for solution.
- 3. Unbalance parameters between <txGain> and <txVol> for receiving, and unbalance parameters between and <rxGain> and <rxVol> for sending.

For example, if one find sending distortion which are not caused by unsuitable FIR parameters and TDD noise, one can adjust <txGain> and <txVol> to a "balance" value. Here "unbalance" means one of <txGain> and <txVol> is set too large, and another is set too small.

5.7 DTMF distortion

Too large sending and receiving gain may result in DTMF distortion.



If one find DTMF sending distortion which can be measured by the oscillograph in receiving end, one can turn off the <micamp1> and try again. Other sending parameters can also be adjusted to get better DTMF performance.

If one find DTMF receiving distortion, on can adjust receiving parameters to get better DTMF performance.

6 Layout guide

The audio signals are sensitive to RF signals and power sources (for example Vbat). Please make sure that the audio signals are far away from the RF signals and Vbat. And the output signals and input signals should be kept away from each other by ground. The differential lines should be layout together. And HPL and HPR are not differential signals, so they should be layout separately.

Filtering capacitors and beads are suggested to be added in the audio lines, 33p and 10p can help reduce the 850Mhz and 1800Mhz RF interfere. If it is signal, the filtering capacitors and beads are suggested to add beside the module pins. If it is output trace, the filtering capacitors and beads are suggested to add beside the handset/headset/speaker connector.

One can send design to us for checking.



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