# **Configure and Test Audio on Raspberry Pi**

# **Hardware Requirements**

- 1. Raspberry Pi Model A/B/B+
- 2. SD Card
- 3. Ethernet Cable / Wi-Fi
- 4. Power Supply
- 5. USB Microphone
- 6. Speakers

# **Software Requirements**

1. Raspbian Stretch OS

# Attach USB Microphone and Speakers to Raspberry Pi.



I am using speakers with audio jack.

1. Select Output device



Advanced Options → Audio → Force 3.5 mm( 'headphone') jack



2. Locate your USB microphone in the list of capture hardware devices. (Write down the card number and device number.) (in my case its 1,0 respectively)

## \$ arecord -l

```
File Edit Tabs Help

pi@raspberrypi:~ $ arecord -1

**** List of CAPTURE Hardware Devices ****

card 1: Device [USB PnP Sound Device], device 0: USB Audio [USB Audio]

Subdevices: 1/1

Subdevice #0: subdevice #0
```

3. Locate your speaker in the list of playback hardware devices.

(Write down the **card number and device number.**)

**Note that** the **3.5mm-jack** is typically labeled **Analog or bcm2835 ALSA** (not bcm2835 IEC958/HDMI). (in my case its **0,0** respectively)

## \$ aplay -l

```
pi@raspberrypi:~ $ aplay -1
**** List of PLAYBACK Hardware Devices ****
card 0: ALSA [bcm2835 ALSA], device 0: bcm2835 ALSA [bcm2835 ALSA]
   Subdevices: 7/7
   Subdevice #0: subdevice #0
   Subdevice #1: subdevice #1
   Subdevice #2: subdevice #2
   Subdevice #3: subdevice #3
   Subdevice #4: subdevice #4
   Subdevice #5: subdevice #5
   Subdevice #6: subdevice #6
card 0: ALSA [bcm2835 ALSA], device 1: bcm2835 ALSA [bcm2835 IEC958/HDMI]
   Subdevices: 1/1
   Subdevice #0: subdevice #0
```

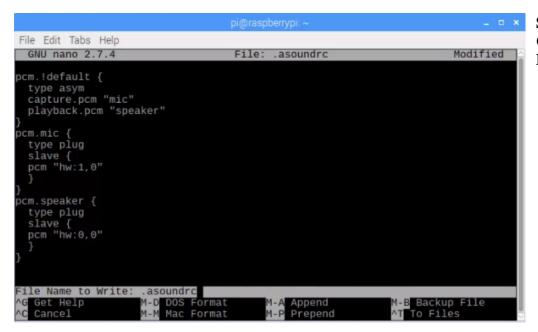
4. Create a new file named .asoundrc in the home directory (/home/pi)

```
pi@raspberrypi:~ _ _ = ×
File Edit Tabs Help
pi@raspberrypi:~ $ pwd
/home/pi
pi@raspberrypi:~ $ nano .asoundrc
```

Add following lines in it

```
pcm.!default {
   type asym
   capture.pcm "mic"
   playback.pcm "speaker"
}
pcm.mic {
   type plug
   slave {
      pcm "hw:<card number>,<device number>"
   }
}
pcm.speaker {
   type plug
   slave {
      pcm "hw:<card number>,<device number>"
}
}
```

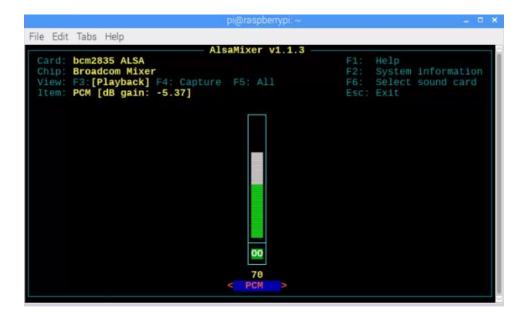
Replace <card number> and <device number> with the numbers you wrote down in the previous step. Do this for both pcm.mic and pcm.speaker.



Save file with Ctrl+O and Enter 5. Adjust the playback volume.

Press the up arrow key to set the playback volume level to around 70.

#### \$ alsamixer



Press Esc to exit.

6. Play a test sound (this will be a person speaking). Press Ctrl+C when done. If you don't hear anything when you run this, check your speaker connection.

### \$ speaker-test -t wav

```
pi@raspberrypi:~ $ speaker-test -t wav
speaker-test 1.1.3
Playback device is default
Stream parameters are 48000Hz, S16_LE, 1 channels
WAV file(s)
Rate set to 48000Hz (requested 48000Hz)
Buffer size range from 512 to 65536
Period size range from 512 to 65536
Using max buffer size 65536
Periods = 4
was set period_size = 16384
was set buffer_size = 65536
0 - Front Left
Time per period = 0.369303
0 - Front Left
Time per period = 1.372140
0 - Front Left
```

7. Record a short audio clip.

\$ arecord --format=S16\_LE --duration=5 --rate=16000 --file-type=raw sample.wav

```
pi@raspberrypi:~ _ _ _ X

File Edit Tabs Help

pi@raspberrypi:~ S arecord --format=S16_LE --duration=5 --rate=16000 --file-type^
=raw sample.wav

Recording raw data 'sample.wav' : Signed 16 bit Little Endian, Rate 16000 Hz, Mo
no
```

8. Check the recording by replaying it

aplay --format=S16\_LE --rate=16000 sample.wav

```
pi@raspberrypi:~ $ aplay --format=S16_LE --rate=16000 sample.wav
Playing raw data 'sample.wav' : Signed 16 bit Little Endian, Rate 16000 Hz, Mono
```

Note that...

- If recording and playback are working, then you are done configuring audio.
- If not, check that the microphone and speaker are properly connected.
- If this is not the issue, then try a different microphone or speaker.

That's all!!!

Thank you....