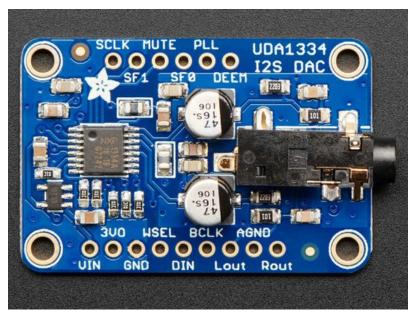


Adafruit I2S Stereo Decoder - UDA1334A

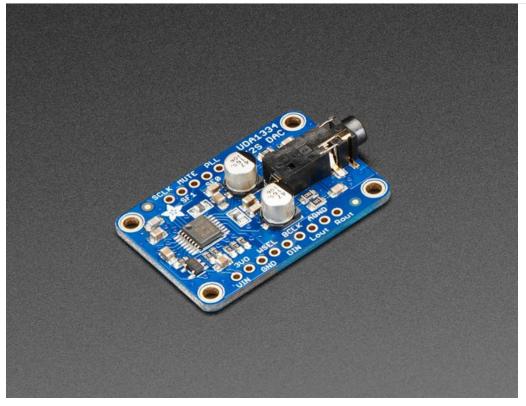
Created by lady ada



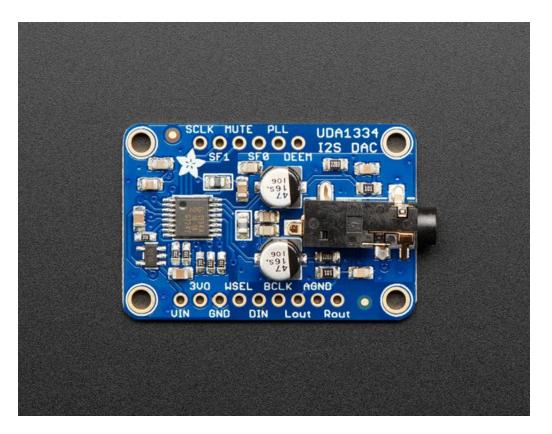
Last updated on 2018-07-20 04:34:40 PM UTC

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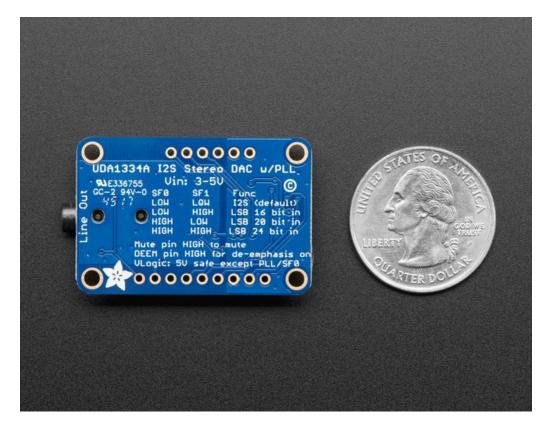
Overview



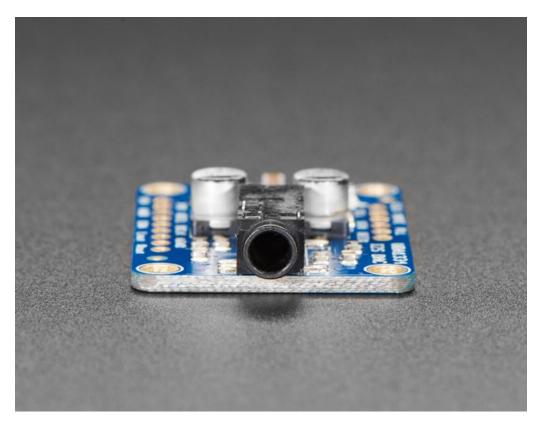
This fully-featured UDA1334A I2S Stereo DAC breakout is a perfect match for any I2S-output audio interface. It's affordable but sounds great! The NXP UDA1334A is a jack-of-all-I2S-trades: you can use 3.3V - 5V logic levels (a rarity), and can process multiple different formats by setting two pins to high or low. The DAC will process data immediately, and give you a clear, analog, stereo line level output. It's even cool with MCLK-less I2S interfaces such as the Raspberry Pi (which it's ideal for) - a built in PLL will generate the proper clock from the incoming signal.



For inputs, you can use classic I2S (the default) or 16-bit, 20-bit or 24-bit left justified data. You can set it up to take an input system/master clock but we default-set it to just generate it for you, so you only need to connect Data In, Word Select (Left/Right Clock) and Bit Clock lines. If you want, there's a mute pin and a de-emphasis filter you can turn on.

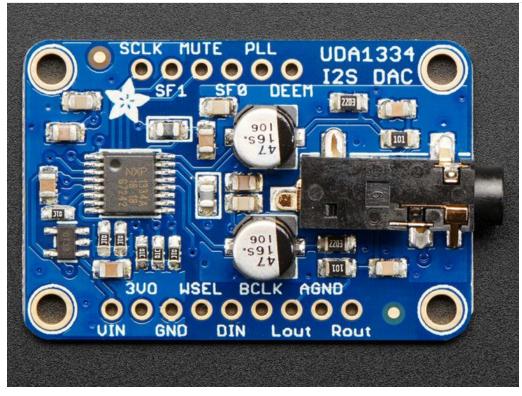


We put in plenty of ferrite beads, a low-dropout regulator, and the recommended band-pass filter so you get a very nice clean output. With a sine-wave generator we swept through 20-20KHz and saw no attenuation or distortion. Plug into either the 3.5mm stereo headphone jack or the breadboard-friendly pads. We think you'll be pleased with this DAC!



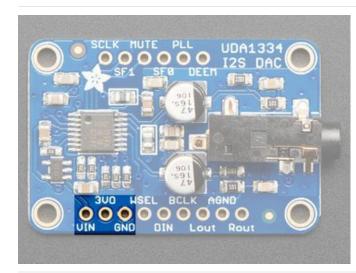
Each order comes with one I2S Stereo DAC breakout and some header you can solder on.

Pinouts



The UDA1334A is an **I2S** amplifier - it does not use analog inputs, it only has digital audio input support! Don't confuse I2S with I2C, I2S is a sound protocol whereas I2C is for small amounts of data.

Power Pins

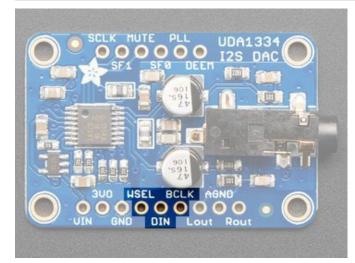


The UDA1334A requires 3.3V power but can take 3-5V level logic on nearly all pins.

You can provide 3-5V power on the **VIN** pin and **GND** and the built in regulator will generate a nice clean 3.3V supplier on **3VO**ut.

Use the quietest power supply for Vin, we do filter the power supply, but the quieter the better!

I2S Pins



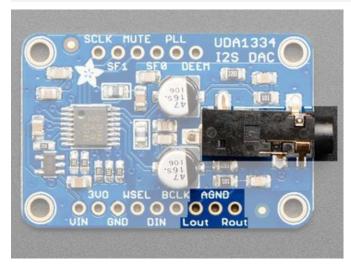
Three pins are used for stereo I2S data in. These pins are required!

These can be 3.3-5V logic

- WSEL (Word Select or Left/Right Clock) this is the pin that tells the DAC when the data is for the left channel and when its for the right channel
- DIN (Data In) This is the pin that has the actual data coming in, both left and right data are sent on this pin, the WSEL pin indicates when left or right is being transmitted
- BCLK (Bit Clock) This is the pin that tells the amplifier when to read data on the data pin.

MCLK is not required to use this DAC, if you have an MCLK pin on your audio source, leave it disconnected.

Audio Outputs



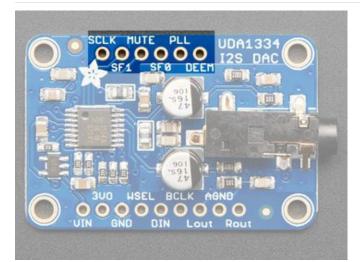
The exciting part! This is where your line level audio comes out. We put big 47uF blocking capacitors on the output so you can connect this to any stereo system. **AGND** is a clean analog ground signal that we recommend using as your analog reference, you'll get a cleaner signal.

Note that this DAC was intended for use with a separate amplifier and is rated for a 3 K Ω load. However, we've found you *can* plug in 32 Ω headphones and the output is current-limited so it won't damage the DAC but you will get distortions. (Powered headphones won't have this issue)

Optional Control Pins

There are some extra configuration pins if you want to use them. They are not required for 99% of usage with an Arduino or Teensy or Raspberry Pi. But you never know! So they are there for you. **PLL** and **SFO** are 3.3V logic only, the other pins are 3-5V safe.

Most of the pins have to do with changing the setup from audio mode to video mode. If you happen to want videomode, for synchronizing with NTSC/PAL, check the datasheet - we haven't used it for that purpose.



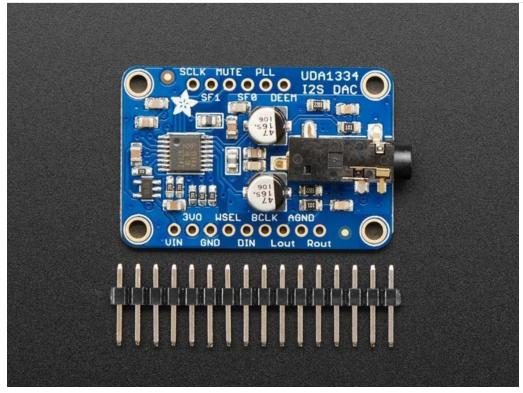
- SCLK (Sys Clock) Optional 27 MHz 'video mode' ssytem clock input - by default we generate the sysclock from the WS clock in 'audio mode' But the UDA can also take a oscillator input on this pin
- Mute Setting this pin High will mute the output
- **De-Em**phasis In audio mode (which is the default), can be used to add a de-emphasis filter. In video mode, where the system clock is generated from an oscillator, this is the clock output.
- PLL sets the PLL mode, by default pulled low for Audio. Can be pulled high or set to ~1.6V to set PAL or NTSC video frequency



SFO and **SF1** are used to set the input data format. By default both are pulled Low for I2S but you can change them around for alternate formats.

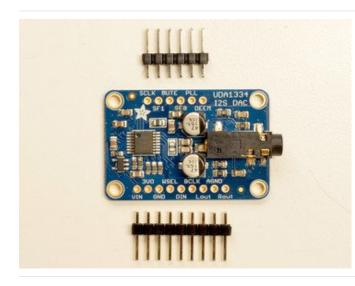
See the back of the PCB for a quick reference

Assembly



Installing Standard Headers

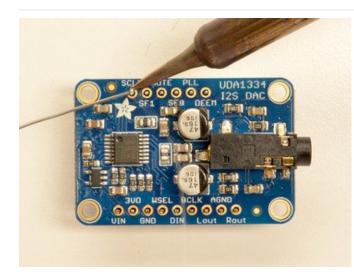
The shield comes with 0.1" standard header.



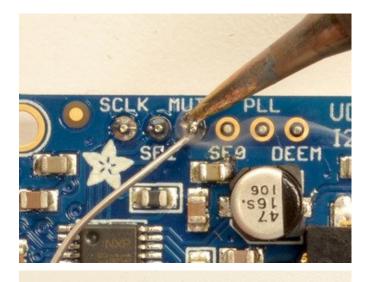
Break apart the 0.1" header into 6 and 9-pin long pieces and slip the short ends into the holes in the board

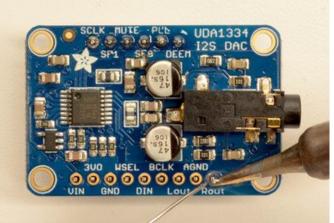


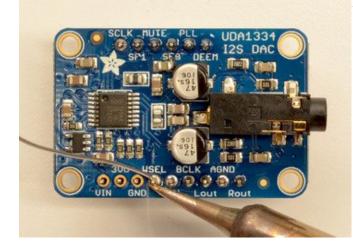
Make sure that all of the short parts of the header are sticking through the two sets of pads on either side of the board

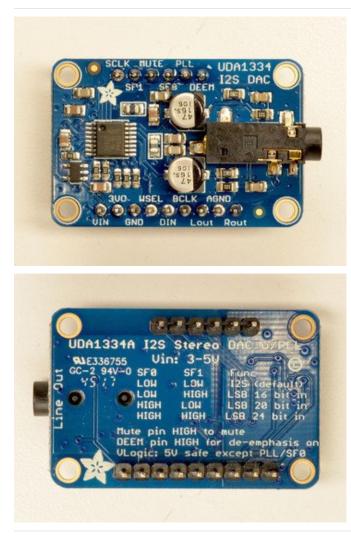


Solder each one of the pins into the board to make a secure connection









That's it! Move on to next page for wiring information

Raspberry Pi Wiring

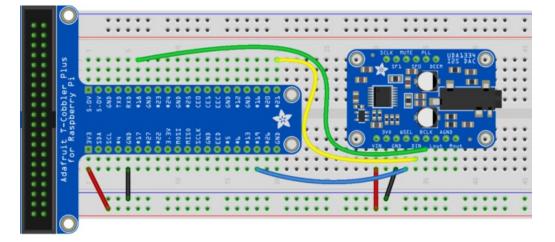
if you have a Raspberry Pi and you want higher quality audio than the headphone jack can provide, I2S is a good option! You only use 3 pins, and since its a pure-digital output, there can be less noise and interference.

This board works very well with boards that *don't* have audio like the Pi Zero and is the easiest way to get quality audio out

This technique will work with any Raspberry Pi with the 2x20 connector. Older Pi 1's with a 2x13 connector do not bring out the I2S pins as easily

Connect:

- Amp Vin to Raspbery Pi 3V or 5V
- Amp GND to Raspbery Pi GND
- Amp DIN to Raspbery Pi #21
- Amp BCLK to Raspbery Pi #18
- Amp LRCLK to Raspbery Pi #19



https://adafru.it/A9T

https://adafru.it/A9T

Raspberry Pi Setup

At this time, Raspbery Pi linux kernel does not support mono audio out of the I2S interface, you can only play stereo, so any mono audio files may need conversion to stereo!

2017-11-2 Raspbian PIXEL ('full') has broken something in volume control. I2S works, but there's no software volume setup, if you need this, try Raspbian Lite - will try to fix as soon as we figure out why :)

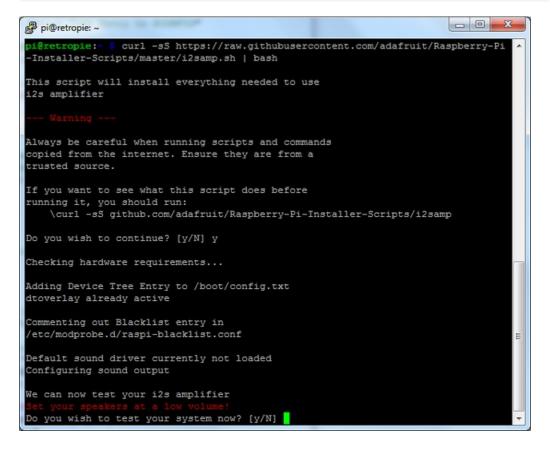
Fast Install

Luckily its quite easy to install support for I2S DACs on Raspbian Jessie.

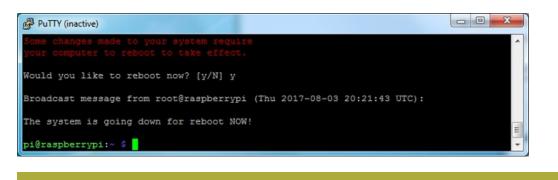
These instructions are totally cribbed from the PhatDAC instructions at the lovely folks at Pimoroni! (https://adafru.it/nFy)

Run the following from your Raspberry Pi with Internet connectivity:

curl -sS https://raw.githubusercontent.com/adafruit/Raspberry-Pi-Installer-Scripts/master/i2samp.sh | bash

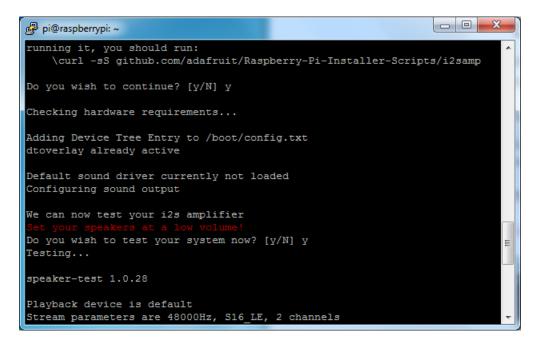


You will need to reboot once installed.



You must reboot to enable the speaker hardware!

After rebooting, log back in and re-run the script again...It will ask you if you want to test the speaker. Say **y**es and listen for audio to come out of your speakers...



In order to have volume control appear in Raspbian desktop or Retropie you must reboot a second time after doing the speaker test, with **sudo reboot**

You can then go to the next page on testing and optimizing your setup. Skip the rest of this page on **Detailed Installation** if the script worked for you!

Detailed Install

If, for some reason, you can't just run the script and you want to go through the install by hand - here's all the steps!

Update /etc/modprobe.d (if it exists)

Log into your Pi and get into a serial console (either via a console cable, the TV console, RXVT, or what have you)

Edit the raspi blacklist with

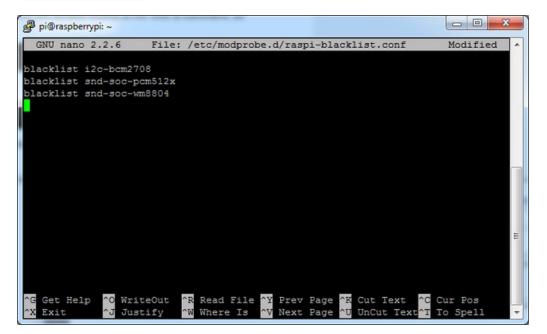
sudo nano /etc/modprobe.d/raspi-blacklist.conf



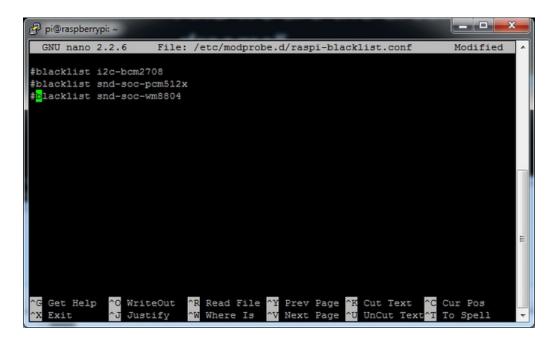
If the file is empty, just skip this step

However, if you see the following lines:

blacklist i2c-bcm2708 blacklist snd-soc-pcm512x blacklist snd-soc-wm8804



Update the lines by putting a # before each line



Save by typing Control-X Y <return>

Disable headphone audio (if it's set)

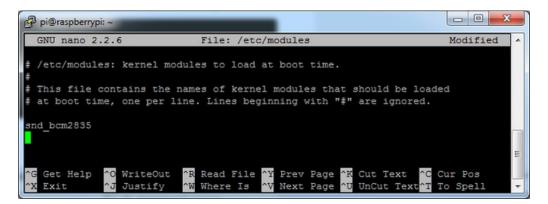
Edit the raspi modules list with

sudo nano /etc/modules

If the file is empty, just skip this step

However, if you see the following line:

snd_bcm2835



Put a # in front of it

³ pi@raspberrypi: ~ GNU nano 2.2.6	File: /etc/modules	Modified
/etc/modules: kernel m	odules to load at boot time.	
	names of kernel modules that shoul line. Lines beginning with "#" are	
and bcm2835		

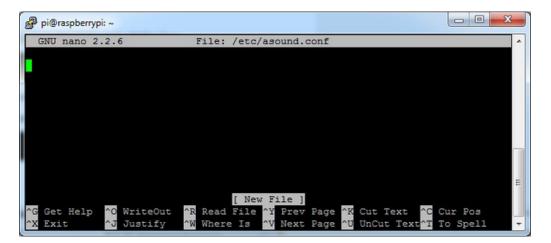
and save with Control-X Y <return>

Create asound.conf file

Edit the raspi modules list with

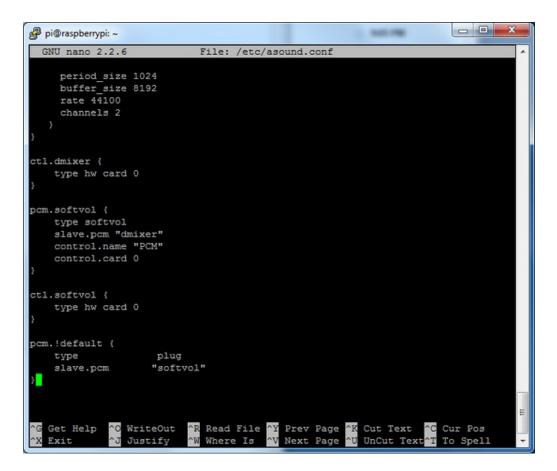
sudo nano /etc/asound.conf

This file ought to be blank!



Copy and paste the following text into the file

```
pcm.speakerbonnet {
   type hw card 0
}
pcm.dmixer {
  type dmix
  ipc key 1024
  ipc_perm 0666
  slave {
    pcm "speakerbonnet"
    period time 0
    period_size 1024
    buffer_size 8192
    rate 44100
    channels 2
  }
}
ctl.dmixer {
   type hw card 0
}
pcm.softvol {
   type softvol
   slave.pcm "dmixer"
   control.name "PCM"
   control.card 0
}
ctl.softvol {
   type hw card 0
}
pcm.!default {
   type
                   plug
   slave.pcm
                 "softvol"
}
```



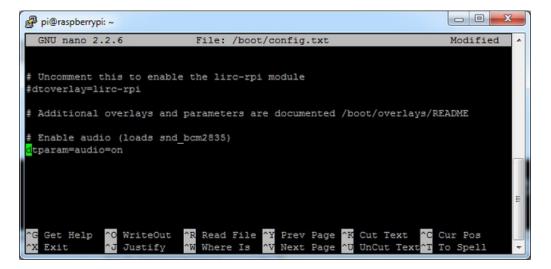
Save the file as usual

Add Device Tree Overlay

Edit your Pi configuration file with

sudo nano /boot/config.txt

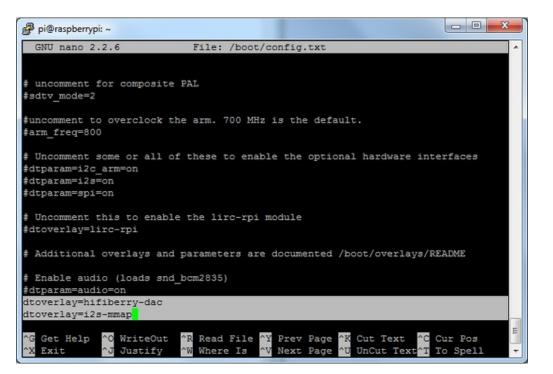
And scroll down to the bottom. If you see a line that says: dtparam=audio=on



Disable it by putting a # in front.

Then add: dtoverlay=hifiberry-dac dtoverlay=i2s-mmap

on the next line. Save the file.



Reboot your Pi with sudo reboot

Raspberry Pi Test Speaker Tests!

OK you can use whatever software you like to play audio but if you'd like to test the speaker output, here's some quick commands that will let you verify your amp and speaker are working as they should!

Simple white noise speaker test

Run speaker-test -c2 to generate white noise out of the speaker, alternating left and right. Since the I2S amp merges left and right channels, you'll hear continuous white noise

Simple WAV speaker test

Once you've got something coming out, try to play an audio file with speaker-test (for WAV files, not MP3)

speaker-test -c2 --test=wav -w /usr/share/sounds/alsa/Front_Center.wav

You'll hear audio coming from left and right alternating speakers

Simple MP3 speaker test

If you want to play a stream of music, you can try

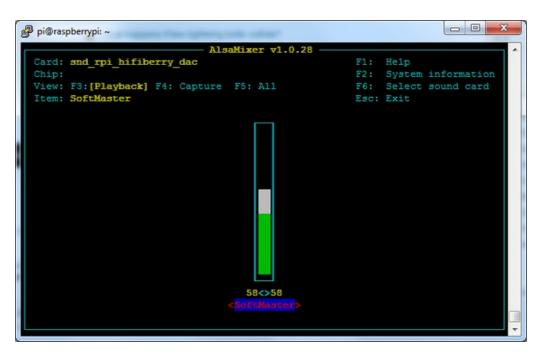
sudo apt-get install -y mpg123 mpg123 http://ice1.somafm.com/u80s-128-mp3

If you want to play MP3's on command, check out this tutorial which covers how to set that up (https://adafru.it/aTD)

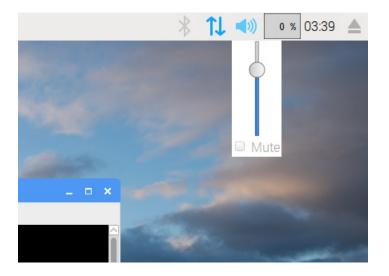
At this time, Jessie Raspbery Pi kernel **does not support mono audio** out of the I2S interface, **you can only play stereo**, so any mono audio files may need conversion to stereo!

Volume adjustment

Many programs like PyGame and Sonic Pi have volume control within the application. For other programs you can set the volume using the command line tool called **alsamixer**. Just type alsamixer in and then use the up/down arrows to set the volume. Press Escape once its set



In Raspbian PIXEL you can set the volume using the menu item control. If it has an X through it, try restarting the Pi (you have to restart twice after install to get PIXEL to recognize the volume control



Pi I2S Tweaks

This page is deprecated, our installer already performs these steps for you, but we'll keep them here for archival use!

Reducing popping

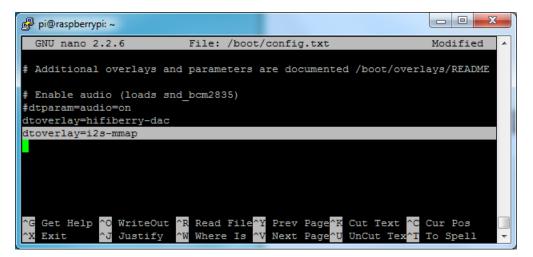
For people who followed our original installation instructions with the simple also config, they may find that the I2S audio pops when playing new audio.

The workaround is to use a software mixer to output a fixed sample rate to the I2S device so the bit clock does not change. I use ALSA so I configured **dmixer** and I no longer have any pops or clicks. Note that the RaspPi I2S driver does not support **dmixer** by default and you must follow these instructions provided (https://adafru.it/sHF) to add it. Continue on for step-by-step on how to enable it!

Step 1

Start by modify /boot/config.txt to add dtoverlay=i2s-mmap

Run sudo nano /boot/config.txt and add the text to the bottom like so:



Save and exit.

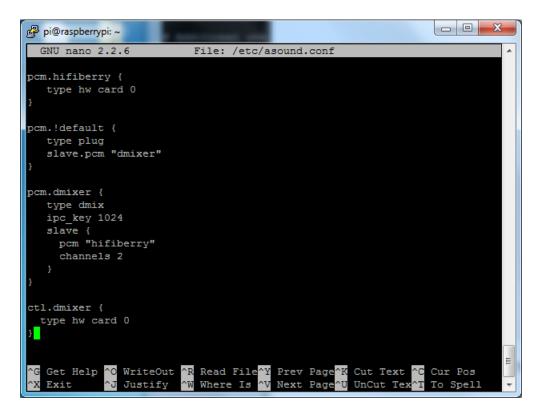
Then change /etc/asound.conf to:

```
pcm.speakerbonnet {
   type hw card 0
}
pcm.!default {
   type plug
   slave.pcm "dmixer"
}
pcm.dmixer {
   type dmix
   ipc key 1024
   ipc_perm 0666
   slave {
    pcm "speakerbonnet"
     period time 0
     period size 1024
     buffer size 8192
     rate 44100
     channels 2
   }
}
ctl.dmixer {
 type hw card 0
}
```

By running sudo nano /etc/asound.conf

This creates a PCM device called speakerbonnet which is connected to the hardware I2S device. Then we make a new 'dmix' device (type dmix) called pcm.dmixer. We give it a unique Inter Process Communication key (ipc_key 1024) and permissions that are world-read-writeable (ipc_perm 0666). The mixer will control the hardware pcm device speakerbonnet (pcm "speakerbonnet") and has a buffer set up so its nice and fast. The communication buffer is set up so there's no delays (period_time 0, period_size 1024) and buffer_size 8192 work well). The default mixed rate is 44.1khz stereo (rate 44100 channels 2)

Finally we set up a control interface but it ended up working best to just put in the hardware device here - ctl.dmixer { type hw card 0 }



Save and exit. Then reboot the Pi to enable the mixer. Also, while it will *greatly* reduce popping, you still may get one once in a while - especially when first playing audio!

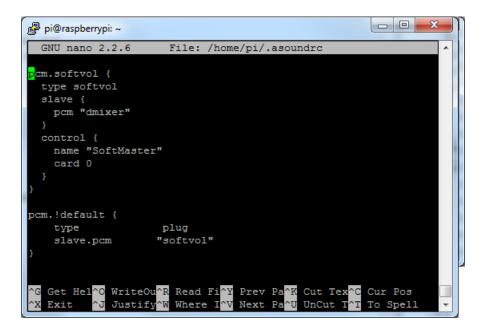
Add software volume control

The basic I2S chipset used here does not have software control built in. So we have to 'trick' the Pi into creating a software volume control. Luckily, its not hard once you know how to do it (https://adafru.it/ydQ).

Create a new audio config file in "/.asoundrc with nano ~/.asoundrc and inside put the following text:

```
pcm.speakerbonnet {
   type hw card 0
}
pcm.dmixer {
  type dmix
  ipc key 1024
  ipc perm 0666
  slave {
    pcm "speakerbonnet"
    period time 0
    period_size 1024
    buffer_size 8192
    rate 44100
    channels 2
  }
}
ctl.dmixer {
   type hw card 0
}
pcm.softvol {
   type softvol
   slave.pcm "dmixer"
   control.name "PCM"
   control.card 0
}
ctl.softvol {
   type hw card 0
}
pcm.!default {
   type
                   plug
   slave.pcm
                 "softvol"
}
```

This assumes you set up the dmixer for no-popping above!



Save and exit

Now, here's the trick, you have to reboot, then play some audio through alsa, then reboot to get the alsamixer to sync up right:

speaker-test -c2 --test=wav -w /usr/share/sounds/alsa/Front_Center.wav

Then you can type alsomixer to control the volume with the 'classic' also mixing interface

Pi@raspberrypi: ~			
AlsaMixer v1.0.28			~
Card: snd_rpi_hifiberry_dac		Help	
Chip:		System information	
View: F3: [Playback] F4: Capture F5: All		Select sound card	
Item: SoftMaster	Esc:	Exit	
58<>58			
< South Management >			
			Ľ

Just press the up and down arrows to set the volume, and ESC to quit

Play Audio with PyGame

You can use **mpg123** for basic testing but it's a little clumsy for use where you want to dynamically change the volume or have an interactive program. For more powerful audio playback we suggest using PyGame to playback a variety of audio formats (MP3 included!)

Install PyGame

Start by installing pygame support, you'll need to open up a console on your Pi with network access and run:

sudo apt-get install python-pygame

Next, download this pygame example zip to your Pi

https://adafru.it/wbp

https://adafru.it/wbp

On the command line, run

wget https://cdn-learn.adafruit.com/assets/assets/000/041/506/original/pygame_example.zip (https://adafru.it/wbq)

unzip pygame_example.zip (https://adafru.it/wbq)

Run Demo

Inside the zip is an example called pygameMP3.py

This example will playback all MP3's within the script's folder. To demonstrate that you can also adjust the volume within pygame, the second argument is the volume for playback. Specify a volume to playback with a command line argument between 0.0 and 1.0

For example here is how to play at 75% volume:

python pygameMP3.py 0.75

Here's the code if you have your own mp3s!

```
''' pg_midi_sound101.py
play midi music files (also mp3 files) using pygame
tested with Python273/331 and pygame192 by vegaseat
'''
#code modified by James DeVito from here: https://www.daniweb.com/programming/software-development/code/4
#!/usr/bin/python
import sys
import pygame as pg
import os
import time
```

```
def play music(music file):
    ...
    stream music with mixer.music module in blocking manner
   this will stream the sound from disk while playing
    ...
   clock = pg.time.Clock()
    try:
       pg.mixer.music.load(music file)
       print("Music file {} loaded!".format(music file))
    except pygame.error:
       print("File {} not found! {}".format(music file, pg.get error()))
       return
    pg.mixer.music.play()
   # If you want to fade in the audio...
    # for x in range(0,100):
          pg.mixer.music.set volume(float(x)/100.0)
   #
         time.sleep(.0075)
    #
   # # check if playback has finished
   while pg.mixer.music.get_busy():
       clock.tick(30)
freq = 44100
              # audio CD quality
bitsize = -16 # unsigned 16 bit
channels = 2
               # 1 is mono, 2 is stereo
buffer = 2048  # number of samples (experiment to get right sound)
pg.mixer.init(freq, bitsize, channels, buffer)
if len(sys.argv) > 1:
   trv:
       user_volume = float(sys.argv[1])
    except ValueError:
       print "Volume argument invalid. Please use a float (0.0 - 1.0)"
       pg.mixer.music.fadeout(1000)
       pg.mixer.music.stop()
       raise SystemExit
    print("Playing at volume: " + str(user volume)+ "\n")
    pg.mixer.music.set volume(user volume)
   mp3s = []
    for file in os.listdir("."):
       if file.endswith(".mp3"):
            mp3s.append(file)
    print mp3s
    for x in mp3s:
       try:
            play music(x)
            time.sleep(.25)
       except KeyboardInterrupt:
            # if user hits Ctrl/C then exit
            # (works only in console mode)
            pg.mixer.music.fadeout(1000)
            pg.mixer.music.stop()
            raise SystemExit
```

```
TALLE OF CONTACT
```

print("Please specify volume as a float! (0.0 - 1.0)")

4

else:

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Arduino Wiring & Test

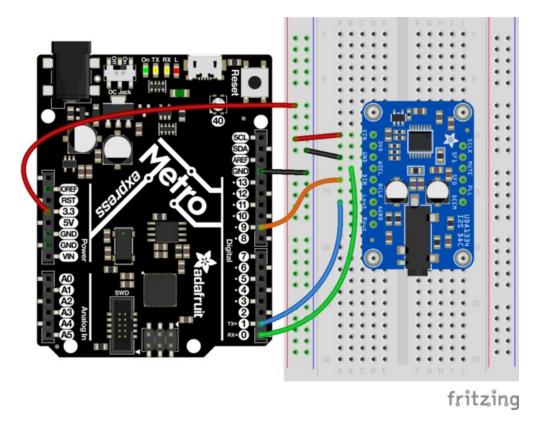
The classic ATmega328P-based Arduino's like the UNO and Metro 328 don't have I2S interfaces, so you *can't* use this breakout with them

But the newer ATSAMD21-based boards like the Zero, Metro M0, Feather M0 can! (Note, Gemma M0 & Trinket M0 do not have I2S pins available).

You may want to check your board documentation to determine where the I2S interface is. We'll use the Metro MO pinout:

- LRCLK / WSEL on Digital #0
- BCLK on Digital #1
- DATA IN on Digital #9

Then power from 3.3V + Ground as usual:



We can then run this demo which will generate a sine wave on both left and right channels. Try tweaking the sample rate, amplitude and frequency, see how the quality and volume changes!

```
/*
This example generates a sine wave based tone at a specified frequency
and sample rate. Then outputs the data using the I2S interface.
Public Domain
*/
#include <I2S.h>
#define FREQUENCY 440 // frequency of sine wave in Hz
#define AMPLITUDE 10000 // amplitude of sine wave
#define SAMPLERATE 44100 // sample rate in Hz
int16 t sinetable[SAMPLERATE / FREQUENCY];
uint32 t sample = 0;
#define PI 3.14159265
void setup() {
 Serial.begin(115200);
 Serial.println("I2S sine wave tone");
 // start I2S at the sample rate with 16-bits per sample
 if (!I2S.begin(I2S_PHILIPS_MODE, SAMPLERATE, 16)) {
   Serial.println("Failed to initialize I2S!");
    while (1); // do nothing
 }
 // fill in sine wave table
  for (uint16 t s=0; s < (SAMPLERATE / FREQUENCY); s++) {</pre>
    sinetable[s] = sin(2.0 * PI * s / (SAMPLERATE/FREQUENCY)) * AMPLITUDE;
 }
}
void loop() {
 if (sample == (SAMPLERATE / FREQUENCY)) {
    sample = 0;
 }
 // write the same sample twice, once for left and once for the right channel
 I2S.write((int16 t) sinetable[sample]); // We'll just have same tone on both!
 I2S.write((int16_t) sinetable[sample]);
 // increment the counter for the next sample in the sine wave table
 sample++;
}
```

CircuitPython Wiring & Test

CircuitPython 3.0 and higher has I2S built in which means you can use this breakout super easily with the supported M0 and M4 Express CircuitPython boards! Supported boards are Feather M0 Express, Feather M4 Express, Metro M0 Express, Metro M4 Express, and ItsyBitsy M0 Express.

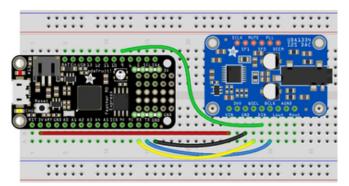
Note that Trinket M0, Gemma M0 and ItsyBitsy M4 do not support I2S (the last one is not a typo!)

The M0 boards have multiple I2S pin combinations available. We're going to demonstrate a single pin combination for each board.

Wiring

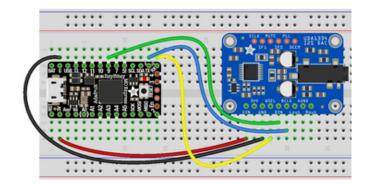
The following wiring diagrams show how to connect the UDA1334 breakout to your CircuitPython board. You'll be using voltage in, ground, bit clock, word select and data pins.

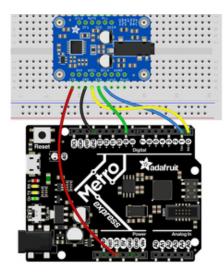
- VIN is the red wire.
- GND is the black wire.
- BCLK is the blue wire.
- WSEL is the yellow wire.
- **DIN** is the **green** wire.

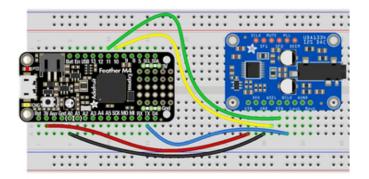


For Feather MO Express, ItsyBitsy MO Express and Metro MO Express:

- Connect VIN on the breakout to 3V/3.3 on the board.
- Connect GND on the breakout to G/GND on the board.
- Connect BCLK on the breakout to D1/TX on the board.
- Connect WSEL on the breakout to DO/RX on the board.
- Connect DIN on the breakout to D9 on the board.

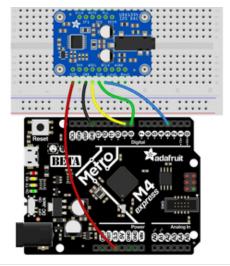






For Feather M4 Express:

- Connect VIN on the breakout to 3V on the board.
- Connect GND on the breakout to Gnd on the board.
- Connect BCLK on the breakout to TX on the board.
- Connect WSEL on the breakout to D10 on the board.
- Connect DIN on the breakout to D11 on the board.



For Metro M4 Express:

- Connect VIN on the breakout to 3.3 on the board.
- Connect GND on the breakout to GND on the board.
- Connect BCLK on the breakout to D3 on the board.
- Connect WSEL on the breakout to D9 on the board.
- Connect DIN on the breakout to D8 on the board.

Code Examples

We have two CircuitPython code examples. The first plays a generated tone through the audio jack on the breakout. The second plays a wave file. Let's take a look!

The default volume of the audio in the following code is very high. Do put on plugged in headpones before first running the code to check the volume.

Tone Generation

The first example generates one period of a sine wave and then loops it to generate a tone. You can change the volume and the Hz of the tone by changing the associated variables. Inside the loop, we play the tone for one second and stop it for one second.

```
import time
import array
import math
import audioio
import board
import audiobusio
tone volume = 0.1 # Increase this to increase the volume of the tone.
frequency = 440 # Set this to the Hz of the tone you want to generate.
length = 8000 // frequency
sine wave = array.array("H", [0] * length)
for i in range(length):
    sine_wave[i] = int((1 + math.sin(math.pi * 2 * i / 18)) * tone_volume * (2 ** 15))
# For Feather M0 Express, ItsyBitsy M0 Express, Metro M0 Express
audio = audiobusio.I2SOut(board.D1, board.D0, board.D9)
# For Feather M4 Express
# audio = audiobusio.I2SOut(board.D1, board.D10, board.D11)
# For Metro M4 Express
# audio = audiobusio.I2SOut(board.D3, board.D9, board.D8)
sine wave sample = audioio.RawSample(sine wave)
while True:
    audio.play(sine wave sample, loop=True)
    time.sleep(1)
    audio.stop()
    time.sleep(1)
```

For Feather MO Express, ItsyBitsy MO Express and Metro MO Express, no changes are needed for the code to work.

For Feather M4 Express, comment out audio = audiobusio.I2SOut(board.D1, board.D0, board.D9) and uncomment # audio = audiobusio.I2SOut(board.D1, board.D1, board.D10, board.D11).

For **Metro M4 Express**, comment out audio = audiobusio.I2SOut(board.D1, board.D0, board.D9) and uncomment # audio = audiobusio.I2SOut(board.D3, board.D3, board.D8).

Now you'll hear one second of a 440Hz tone, and one second of silence. Remember, listen for it without headphones on your ears first as the volume is quite high.

You can try changing the Hz of the tone to produce different tones. Try changing the number of seconds in time.sleep() to produce longer or shorter tones.

Wave File

The second example plays a wave file. We open the file in a readable format. Then inside the loop, we play the file and tell the code to continue playing the file until it's completed. You can use any supported wave file (https://adafru.it/BRj). We've included the wave file used in the code.

https://adafru.it/BTM

https://adafru.it/BTM

```
import audioio
import board
import audiobusio
wave_file = open("StreetChicken.wav", "rb")
wave = audioio.WaveFile(wave_file)
# For Feather M0 Express, ItsyBitsy M0 Express, Metro M0 Express
audio = audiobusio.I2SOut(board.D1, board.D0, board.D9)
# For Feather M4 Express
# audio = audiobusio.I2SOut(board.D1, board.D10, board.D11)
# For Metro M4 Express
# audio = audiobusio.I2SOut(board.D3, board.D9, board.D8)
while True:
    audio.play(wave)
    while audio.playing:
        pass
```

The object setup in the code is the same as above.

For Feather MO Express, ItsyBitsy MO Express and Metro MO Express, no changes are needed for the code to work.

For Feather M4 Express, comment out audio = audiobusio.I2SOut(board.D1, board.D0, board.D9) and uncomment # audio = audiobusio.I2SOut(board.D1, board.D10, board.D11).

For **Metro M4 Express**, comment out audio = audiobusio.I2SOut(board.D1, board.D0, board.D9) and uncomment # audio = audiobusio.I2SOut(board.D3, board.D3, board.D8).

Now you'll hear the wave file play through and loop. Remember, listen for it without headphones on your ears first as the volume is quite high.

There's plenty you can do with this example. Try playing a different wave file, or, instead of including while audio.playing: pass, include a time.sleep() to have it play for a specified number of seconds. Check out the Audio Out page in the CircuitPython Essentials guide (https://adafru.it/BRj) for pause and resume features.

For more information about I2SOut, check out <u>https://circuitpython.readthedocs.io/en/latest/shared-bindings/audiobusio/I2SOut.html</u>

Where's my I2S?

We mentioned earlier that the supported M0 boards have multiple I2S pin combinations available to you. The M4 boards have one option. Either way, if you'd like to know what options are available to you, copy the following code into your **code.py**, **connect to the serial console**, and **check out the output**.

These are the results from the ItsyBitsy MO Express.

```
Adafruit CircuitPython 3.0.0 on 2018-07-09; Adafruit ItsyBitsy M0 Express with samd21g18
>>>
soft reboot
Auto-reload is on. Simply save files over USB to run them or enter REPL to disable.
code.py output:
Bit clock pin: board.D1
                                 Word select pin: board.D0
                                                                 Data pin: board.D4
                                 Word select pin: board.D0
                                                                 Data pin: board.D9
Bit clock pin: board.D1
Bit clock pin: board.D1
                                 Word select pin: board.D0
                                                                 Data pin: board.D12
Bit clock pin: board.D1
                                 Word select pin: board.D7
                                                                 Data pin: board.D4
Bit clock pin: board.D1
                                 Word select pin: board.D7
                                                                 Data pin: board.D9
Bit clock pin: board.D1
                                 Word select pin: board.D7
                                                                 Data pin: board.D12
Bit clock pin: board.D6
                                 Word select pin: board.D0
                                                                 Data pin: board.D4
Bit clock pin: board.D6
                                 Word select pin: board.D0
                                                                 Data pin: board.D9
Bit clock pin: board.D6
                                 Word select pin: board.DO
                                                                 Data pin: board.D12
Bit clock pin: board.D6
Bit clock pin: board.D6
Bit clock pin: board.D6
                                 Word select pin: board.D7
                                                                 Data pin: board.D4
                                                                 Data pin: board.D9
                                 Word select pin: board.D7
                                                                 Data pin: board.D12
                                 Word select pin: board.D7
 import board
 import audiobusio
 from microcontroller import Pin
 def is hardware i2s(bit clock, word select, data):
     try:
          p = audiobusio.I2SOut(bit clock, word select, data)
         p.deinit()
         return True
     except ValueError:
          return False
 def get unique pins():
     exclude = ['NEOPIXEL', 'APA102 MOSI', 'APA102 SCK']
     pins = [pin for pin in [
         getattr(board, p) for p in dir(board) if p not in exclude]
              if isinstance(pin, Pin)]
     unique = []
     for p in pins:
          if p not in unique:
              unique.append(p)
     return unique
 for bit clock pin in get unique pins():
     for word select pin in get unique pins():
          for data pin in get unique pins():
              if bit clock pin is word select pin or bit clock pin is data pin or word select pin\
                       is data pin:
                  continue
              else:
                  if is hardware i2s(bit clock pin, word select pin, data pin):
                       print("Bit clock pin:", bit_clock_pin, "\t Word select pin:", word_select_pin,
                              "\t Data pin:", data pin)
                  else:
                       pass
```

Downloads

Files

- EagleCAD PCB Files (https://adafru.it/BtN)
- Fritzing object in Adafruit Fritzing library (https://adafru.it/aP3)
- UDA1334A Datasheet (https://adafru.it/BtO)

Schematic & Fabrication Print

