

C++ Real-Time Audio Programming with Bela

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Course topics

Programming topics

- Working in real time
- Buffers and arrays
- Parameter control
- Classes and objects
- Analog and digital I/O
- Filtering
- Circular buffers
- Timing in real time
- State machines
- MIDI
- Block-based processing
- Threads
- Fixed point arithmetic
- ARM assembly language



Music/audio topics

- Oscillators
- Samples
- Wavetables
- Filters
- Control voltages
- Gates and triggers
- Delays and delay-based effects
- Metronomes and clocks
- Envelopes
- ADSR
- MIDI
- Additive synthesis
- Phase vocoders
- Impulse reverb

Today

Lecture 6: Analog I/O

What you'll learn today:

Analog inputs and outputs on Bela
Working with breadboards
Basics of control voltages

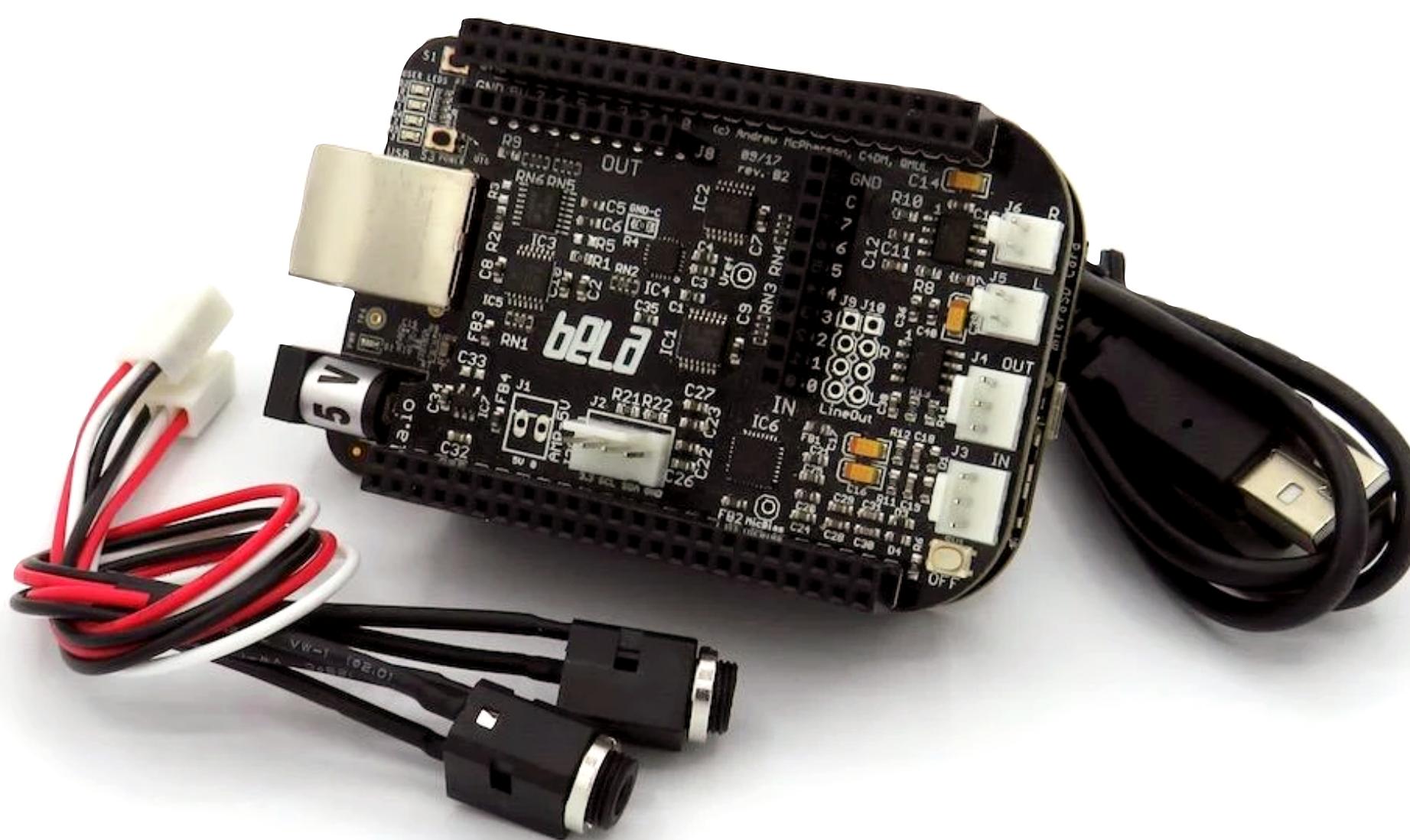
What you'll make today:

A VCO and a simple musical instrument

Companion materials:

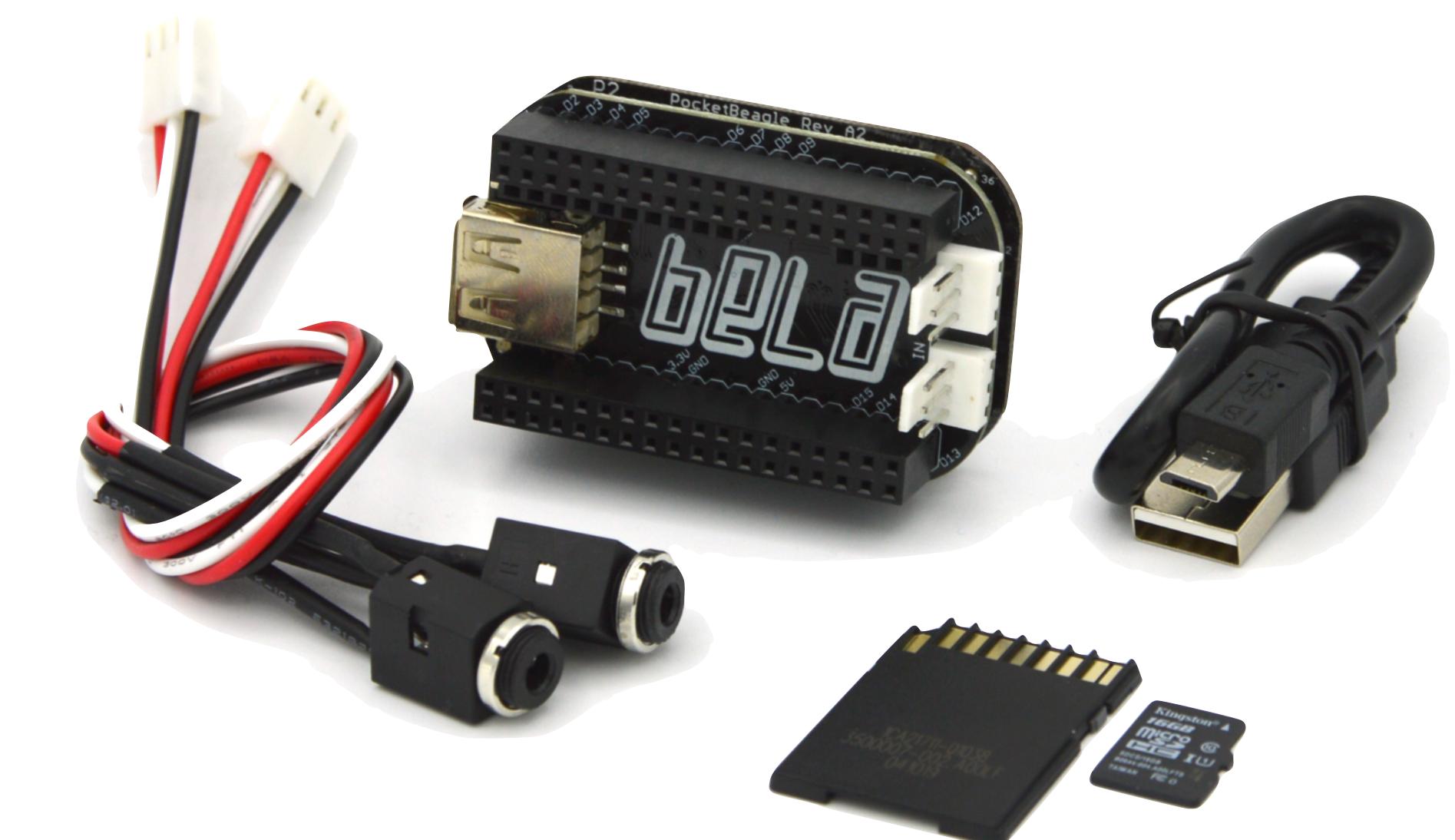
github.com/BelaPlatform/bela-online-course

What you'll need



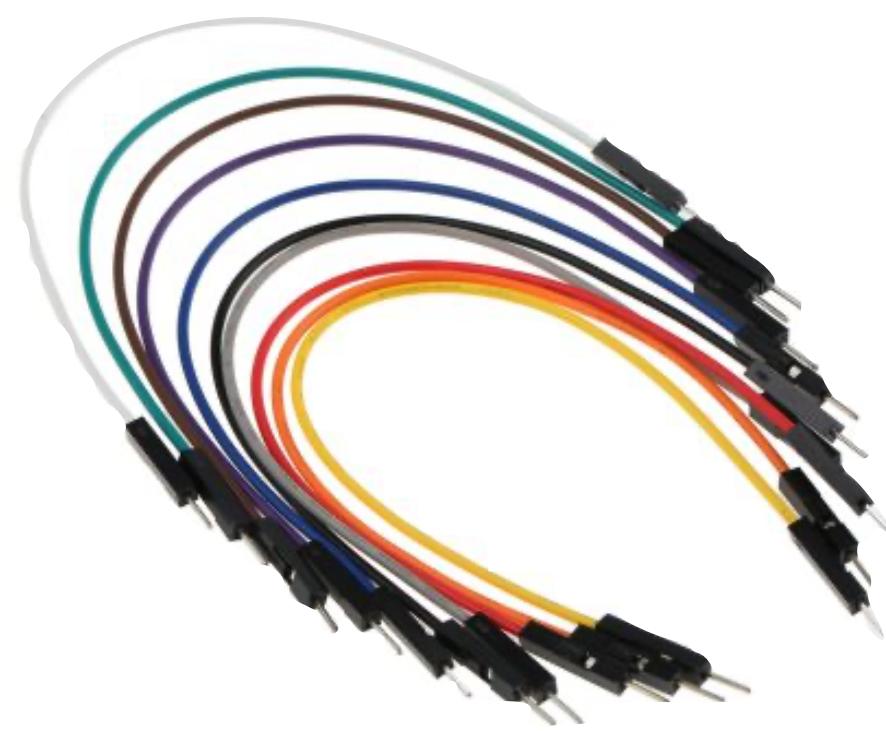
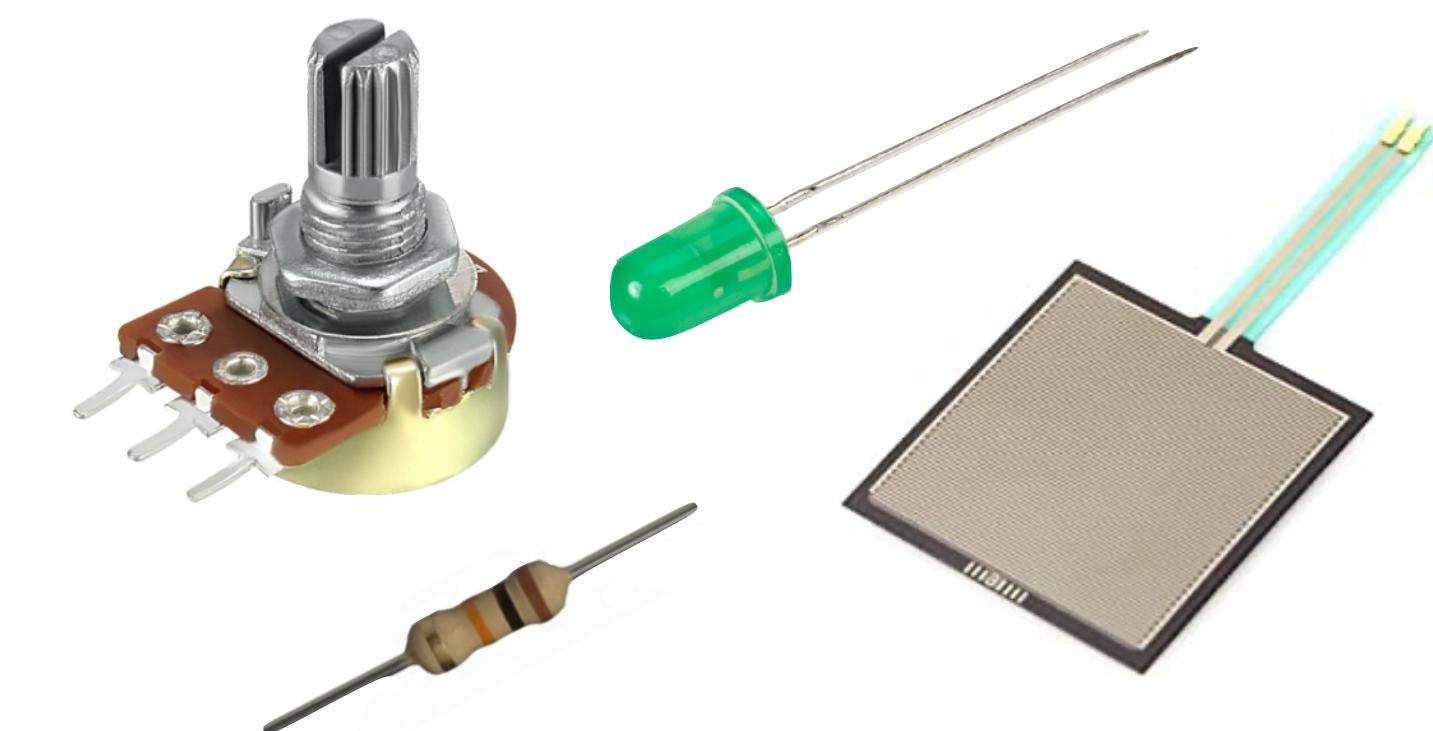
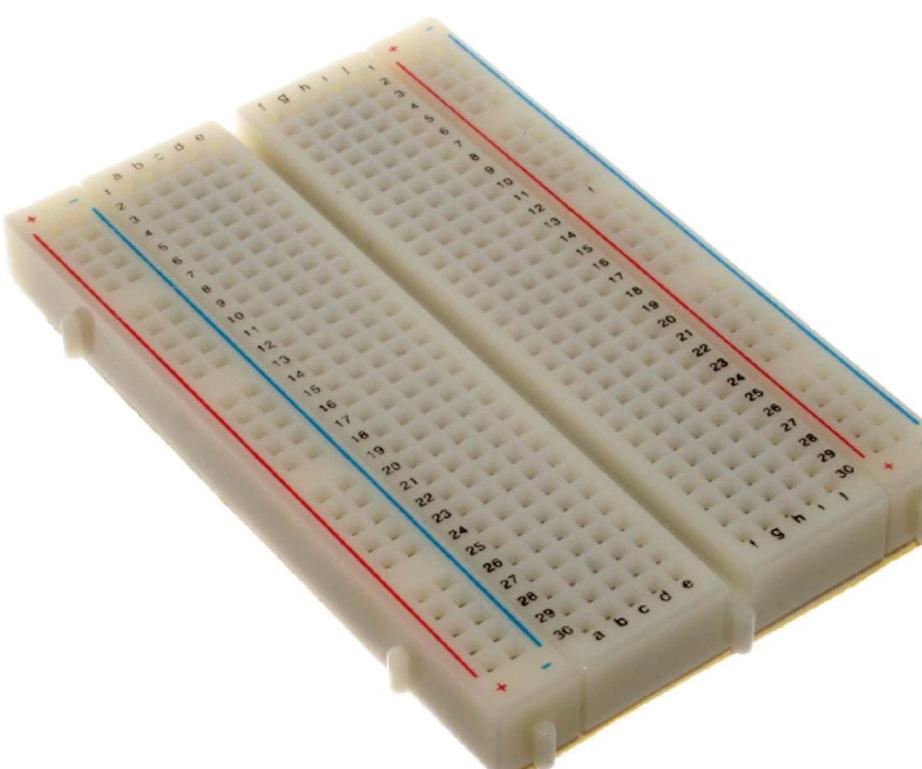
Bela Starter Kit

or

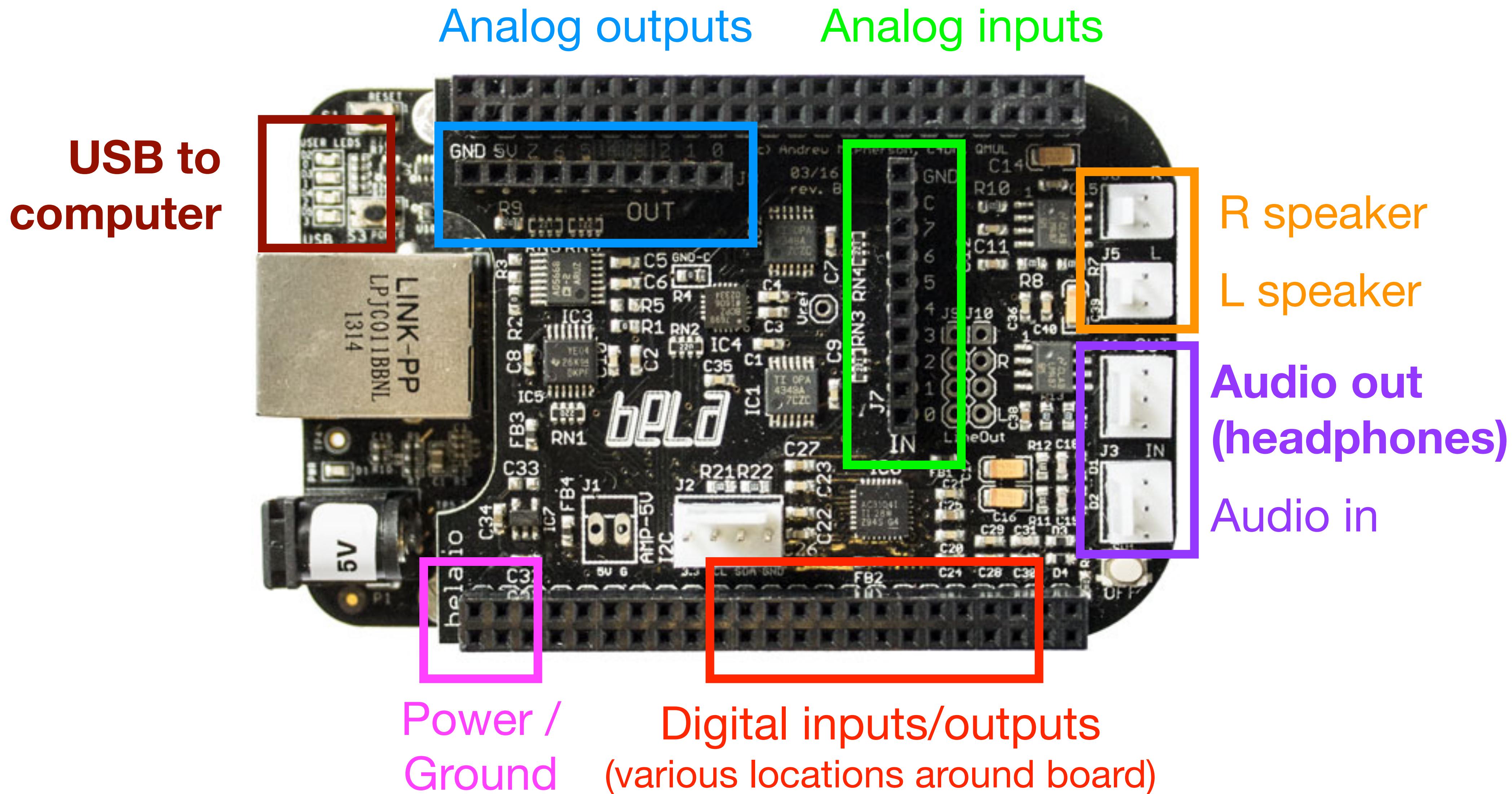


Bela Mini Starter Kit

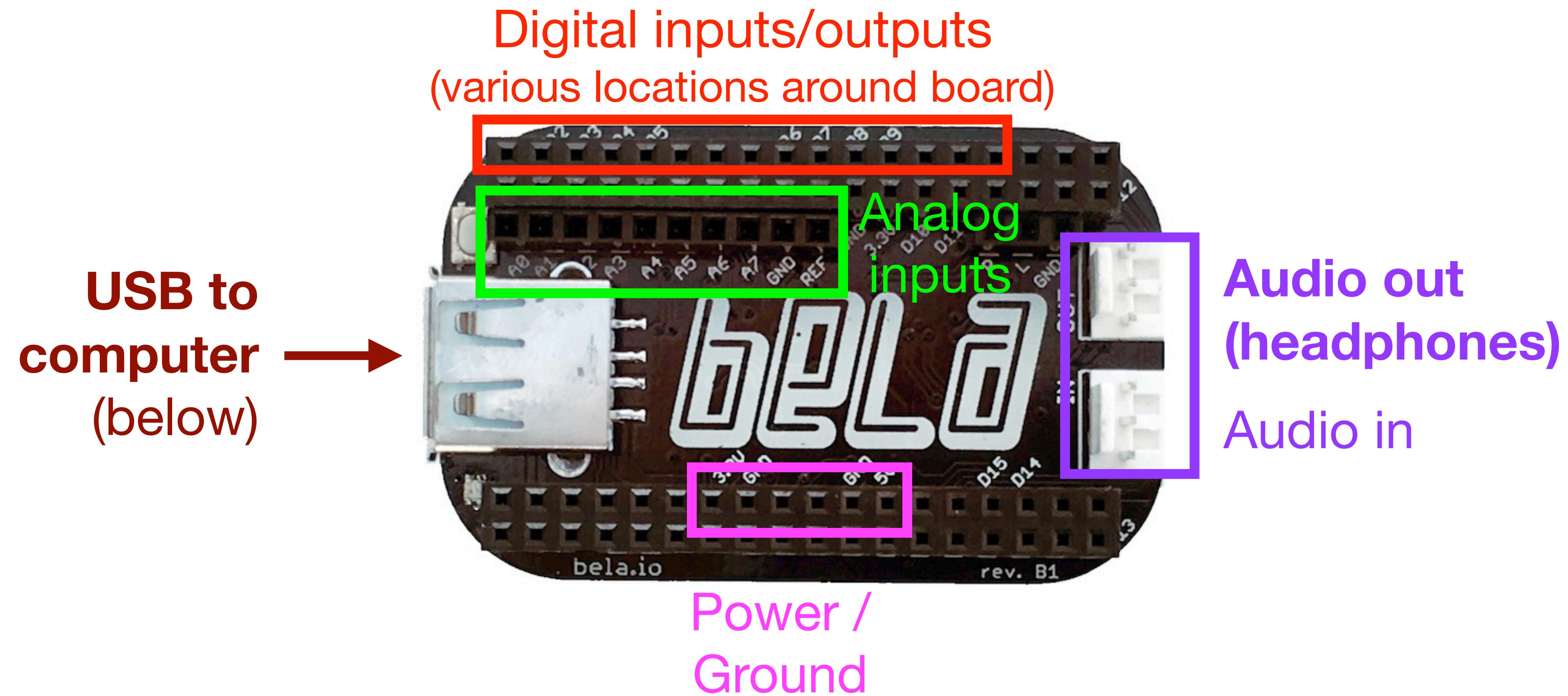
Also needed for
this lecture:



Bela hardware

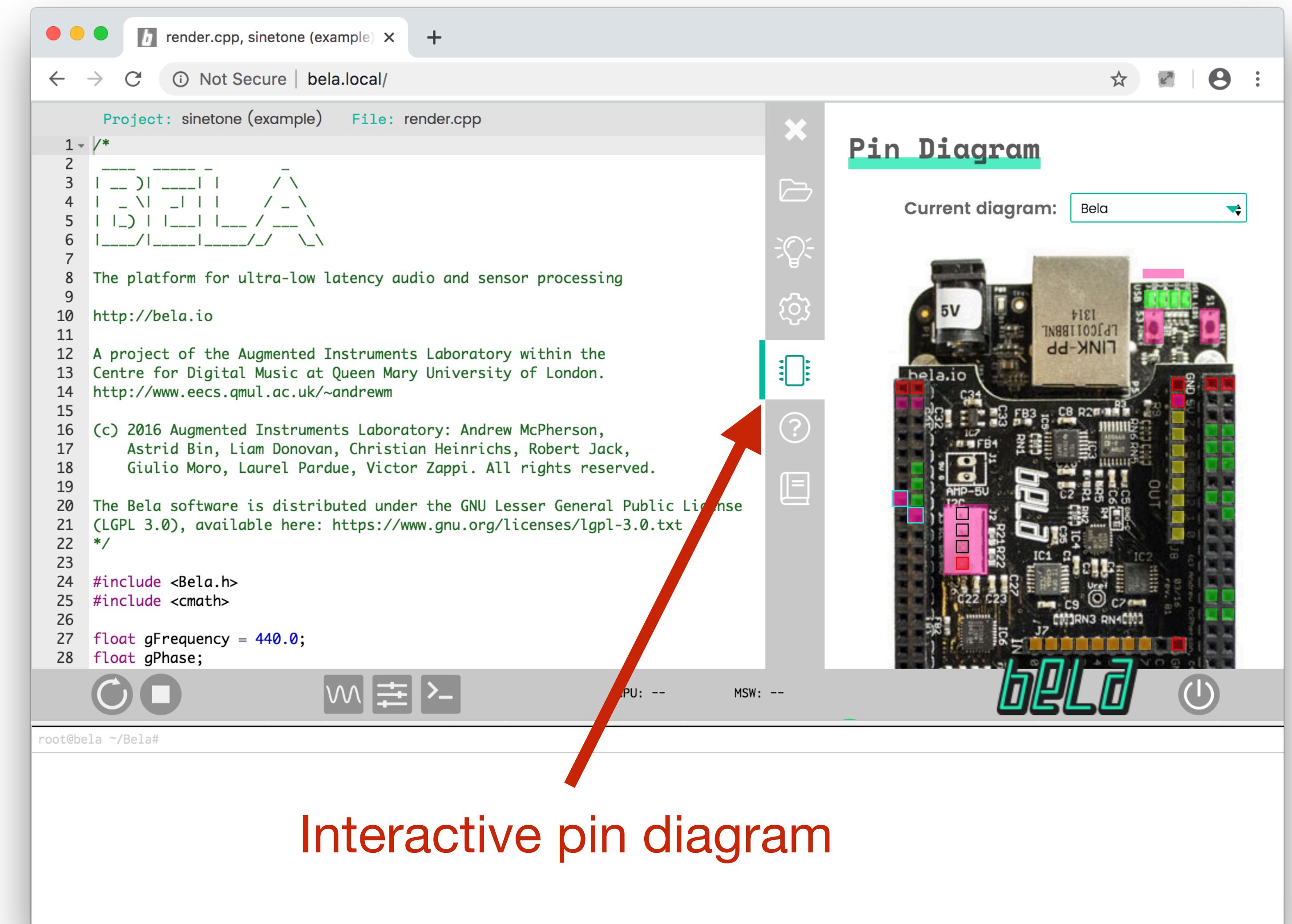


Bela Mini hardware



Analog input

- You can always find the pinout for the analog input and other connectors in the IDE's **Interactive Pin Diagram**



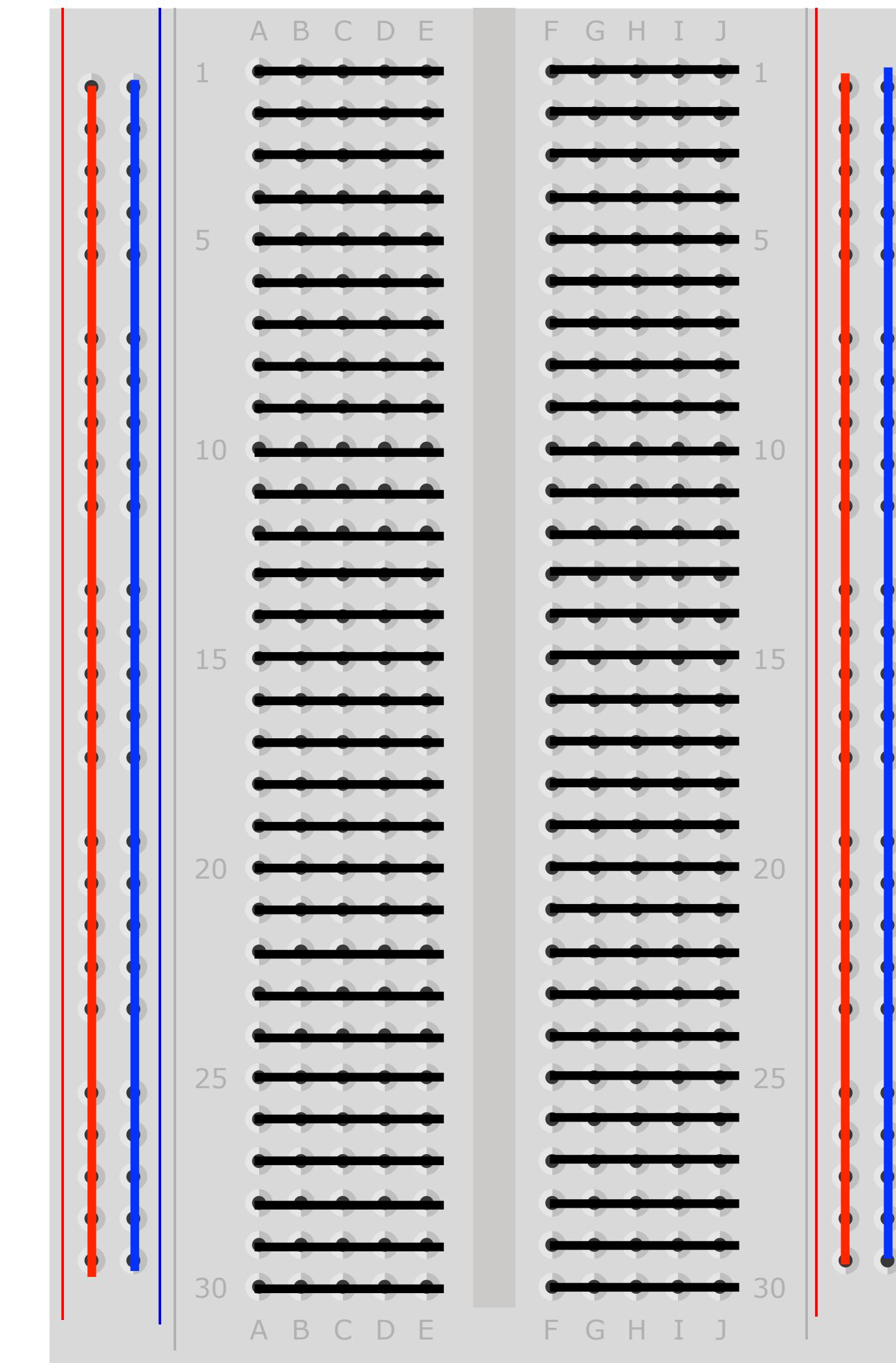
Using a breadboard

Each column of 5 holes
connects together...

...but not across the
break in the middle.

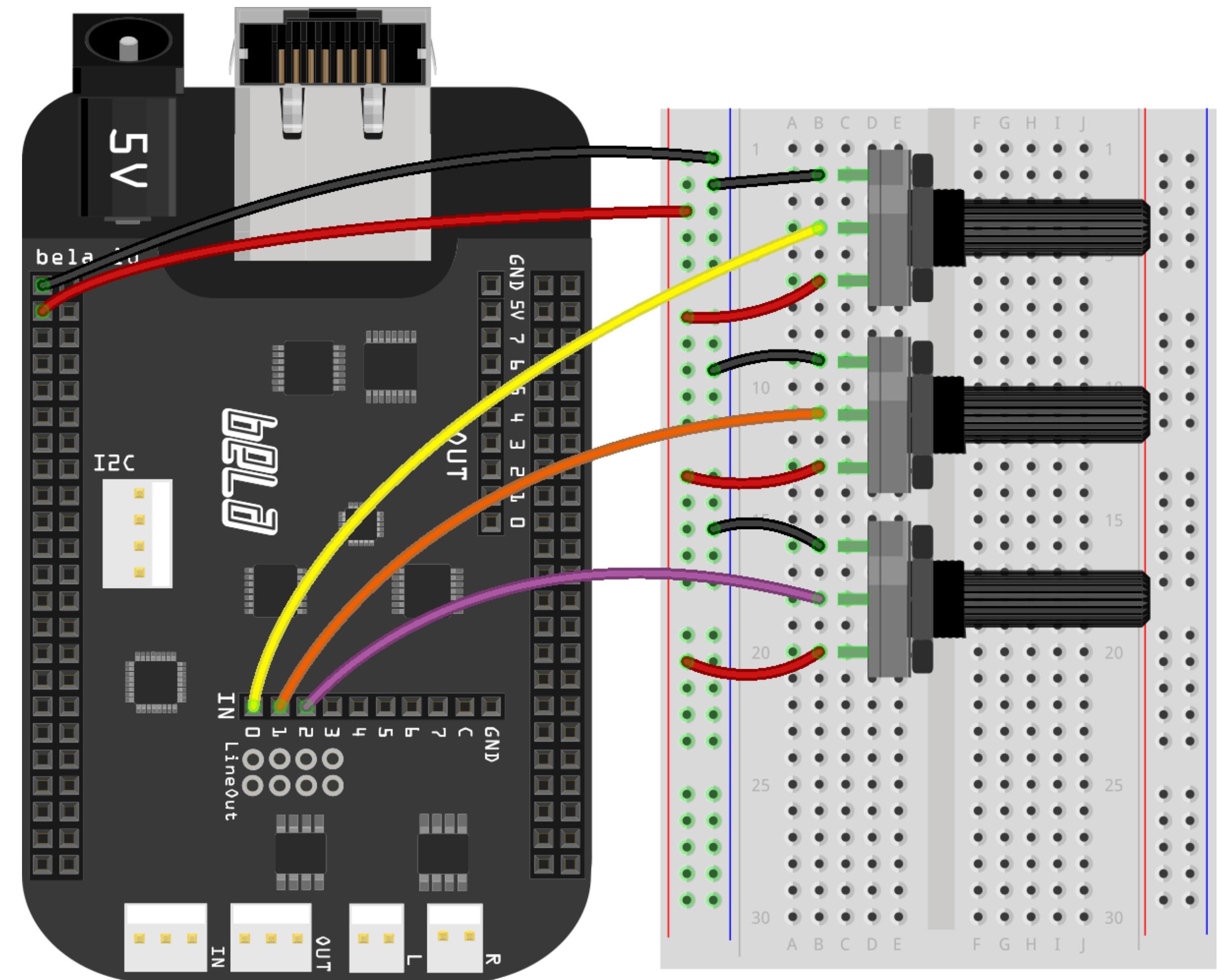
Long rows at the ends
each connect together.

Typically used for
power and **ground**



Connect two potentiometers

- Bela has sources of 3.3V and 5V on board
- We will **always use 3.3V** in this course as 5V **will** damage the digital inputs!
- Wire up the circuit on the right and open the scope-analog example in the Examples tab of the Bela IDE



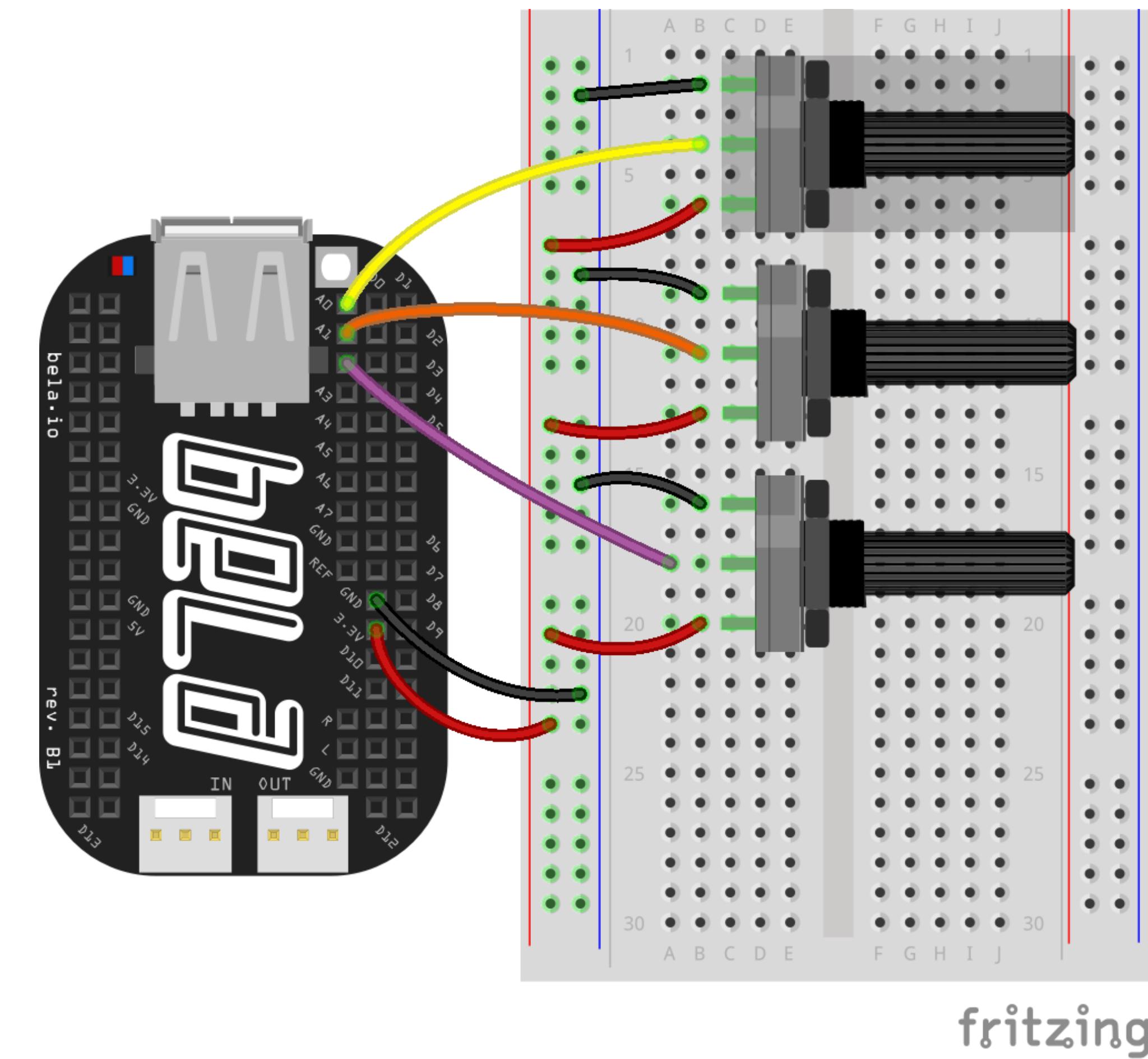
fritzing

Ground
(P9_1) and
+3.3V
(P9_3) to
long rows

Middle wires of
potentiometers
go to Analog In 0
and Analog In 1

Connect two potentiometers

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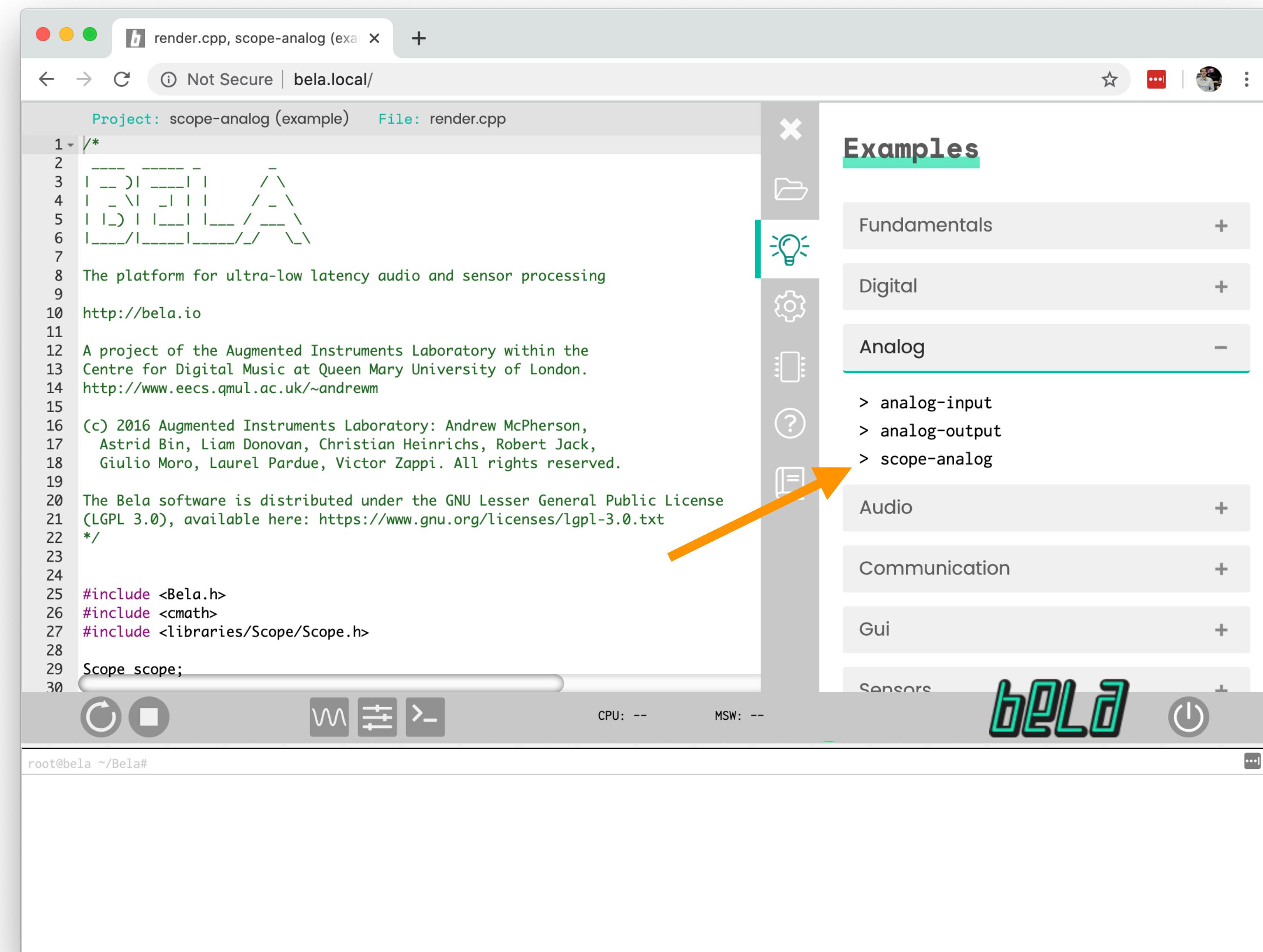


Ground
(P9_1) and
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Middle wires of
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go to **Analog In 0**
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Working with analog data

- Analog inputs on Bela are **sampled signals**, just like audio
 - You don't have to ask the Bela hardware to take the readings; they are **captured automatically in a buffer** at a constant sample rate
 - Thus, we automatically have access to signals representing **voltage** versus time

Voltage

Analog inputs are **DC-coupled**

16-bit resolution

Usable input range 0-4.096V

Software values 0-1 (as a float)



0V corresponds to a reading of **0.0**;
4.096V corresponds to a reading of **1.0**

- What does a reading of **0.5** correspond to in voltage?
2.048V
- What is the expected reading for a **3.3V** input?
 $3.3V / 4.096V \approx 0.806$

Working with analog data

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Usable input range **0-4.096V**

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Time

Sample rate depends on the
number of input channels:

8 channels
(default)

22.05kHz
(half audio rate)

4 channels

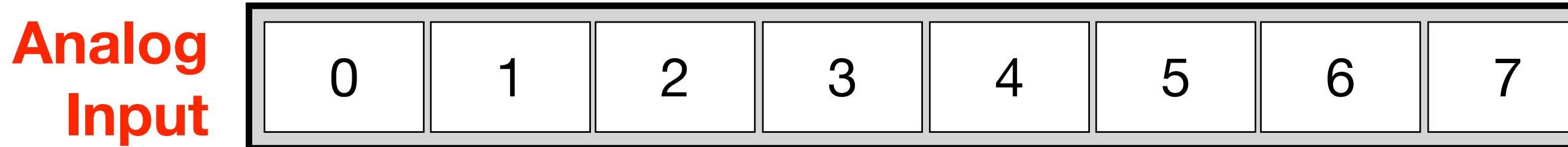
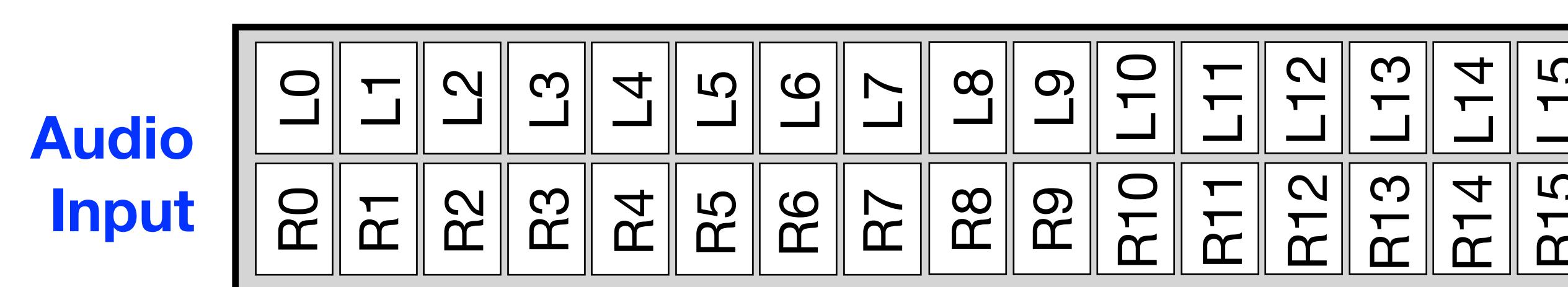
44.1kHz
(audio rate)

2 channels

88.2kHz
(double audio rate)

Analog and audio frames

- The default analog input sample rate is 22.05kHz
 - Audio is sampled at 44.1kHz. That's a period of 22.7 μ s.
 - What is the sampling period of the analog input? 45.4 μ s
 - At this sample rate, how many audio frames elapse for each analog frame? 2
- Each real-time block covers a particular slice of time
 - If there were 16 audio frames in the block, how many analog frames?



There are 8 input channels like this, 8 frames each

$\frac{1}{2}$
1 analog period
= 2 audio periods

Given a particular audio frame number, how do we find the corresponding analog frame?

divide by 2

Why doesn't this result in a fractional index?

we use int for indexing

Analog API

- Bela analog buffers are interleaved, holding data that is already sampled
- `analogRead()` is a convenience function that accesses the analog buffer
 - ▶ Pulls out the particular frame and channel we want

```
float analogRead(BelaContext *context, int frame, int channel) {  
    return context->analogIn[frame * context->analogInChannels + channel];  
}
```

Reading
is in the
return value

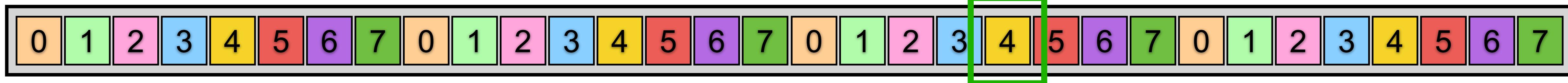
Reference to the `BelaContext`
structure, which holds the
actual data we want

Which frame
(i.e. `sample`) within the
buffer to read

Which `channel`
(e.g. 0, 1, ..., 7)
to read

For example: `float reading = analogRead(context, 2, 4);`

`context->analogIn`



Audio frames and analog frames

- Here is the scope-analog code for reading the analog inputs in `render()`:

```
for(unsigned int n = 0; n < context->audioFrames; n++) {  
  
    if(gAudioFramesPerAnalogFrame && !(n % gAudioFramesPerAnalogFrame))  
        // Read analog inputs and update frequency and amplitude  
        gIn1 = analogRead(context, n/gAudioFramesPerAnalogFrame, 0); ←  
        gIn2 = analogRead(context, n/gAudioFramesPerAnalogFrame, 1); ←  
        gAmplitude = gIn1 * 0.8f;  
        gFrequency = map(gIn2, 0, 1, 100, 1000); ←  
    }  
  
    float out = gAmplitude * sinf(gPhase);  
    // ...  
}
```

??? ← `analogRead()` takes the frame and channel and gets a value from the BelaContext

map() rescales the range from 0-1 to 100-1000 (Hz)

- The variable `gAudioFramesPerAnalogFrame` is calculated in `setup()`:

```
if(context->analogFrames)  
    gAudioFramesPerAnalogFrame = context->audioFrames / context->analogFrames;
```

Audio frames and analog frames

- Suppose we knew the analog sample rate was half the audio rate
 - ▶ We could write:

Modulo (%) means
remainder after
division. Is n even?

```
for(unsigned int n = 0; n < context->audioFrames; n++) {  
  
    if((n % 2) == 0) { // Read analog on the even audio samples  
        // Read analog inputs and update frequency and amplitude  
        gIn1 = analogRead(context, n/2, 0);  
        gIn2 = analogRead(context, n/2, 1);  
        gAmplitude = gIn1 * 0.8f;  
        gFrequency = map(gIn2, 0, 1, 100, 1000);  
  
        float out = gAmplitude * sinf(gPhase);  
        // ...  
    }  
}
```

Convert **audio frame** number
(n) to **analog frame** number (n/2)

Audio frames and analog frames

- Suppose we knew the analog sample rate was equal to the audio rate
 - ▶ We could write:

```
for(unsigned int n = 0; n < context->audioFrames; n++) {  
  
    // Read analog inputs and update frequency and amplitude  
    gIn1 = analogRead(context, n, 0);  
    gIn2 = analogRead(context, n, 1);  
    gAmplitude = gIn1 * 0.8f;  
    gFrequency = map(gIn2, 0, 1, 100, 1000);  
  
    float out = gAmplitude * sinf(gPhase);  
    // ...  
}
```

Even simpler!

- We use `gAudioFramesPerAnalogFrame` because we don't know the rate

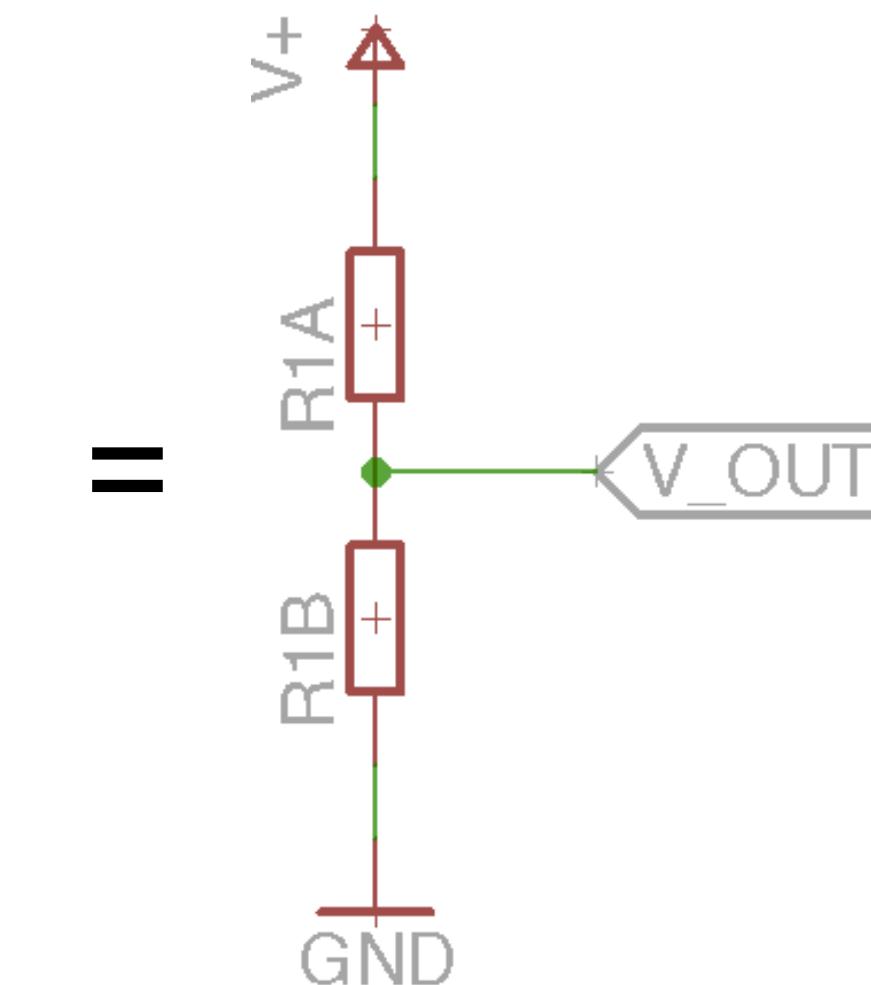
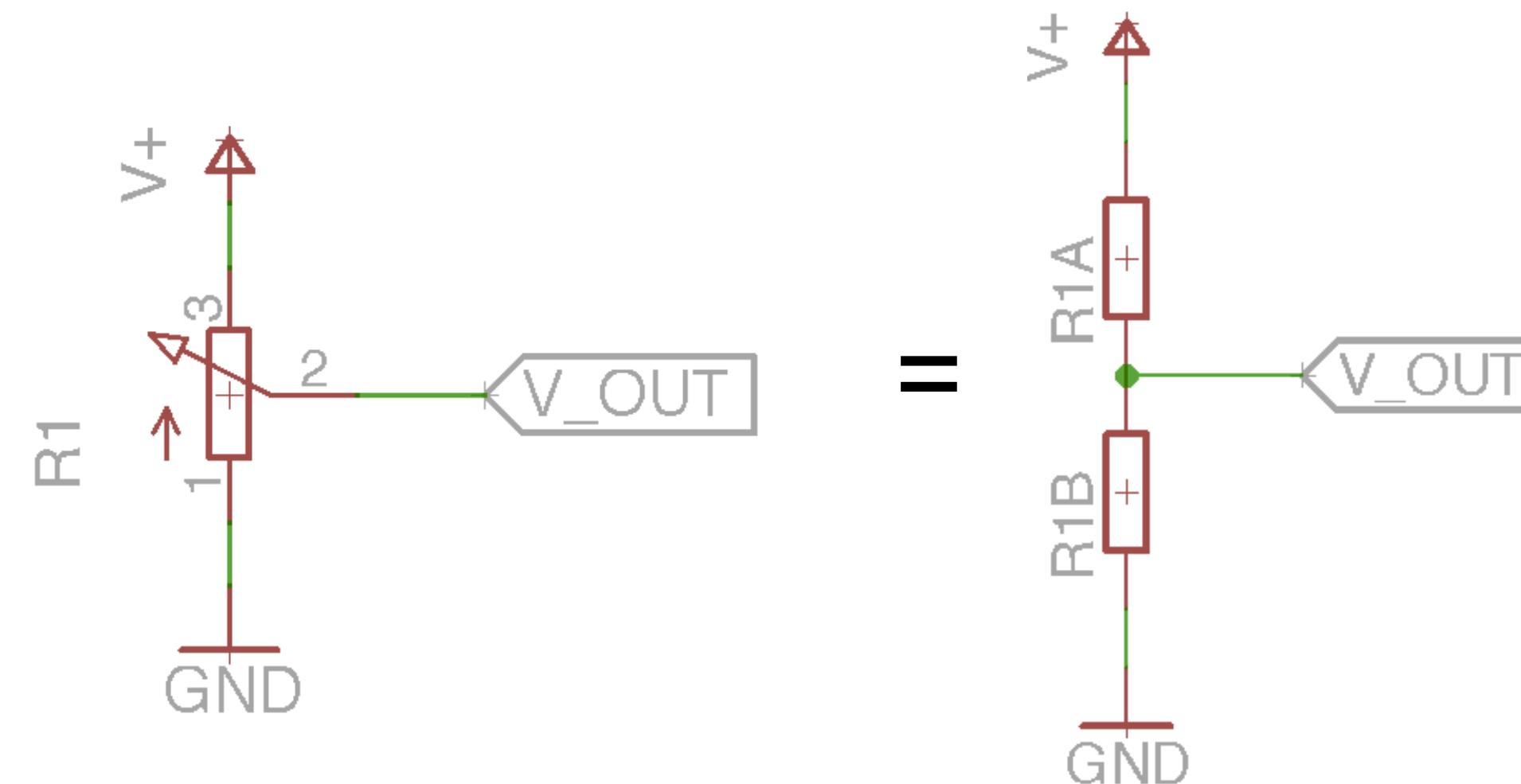
```
if(gAudioFramesPerAnalogFrame && !(n % gAudioFramesPerAnalogFrame)) { }
```

Voltage-controlled oscillator

- We have been making a VCO (voltage-controlled oscillator)
 - This is a standard component of analog synths
 - The control voltage (CV) input changes the frequency of the oscillator
 - The analog input is what reads the control voltage
- A common standard in the synth industry is 1V/octave
 - Every additional 1V on the CV produces a doubling of the frequency
 - On Bela, analog input takes a range of 0 to 4.096V, so what is 1V? $1 / 4.096 = .2441$
- Task: using the vco example,
 - Change the frequency, amplitude and detune controls from GUI sliders to analog inputs
 - Then, make the control a 1V/octave range, with a minimum frequency of 55Hz
 - First rescale the analog input to volts, then implement the formula: $f_{out} = f_{ref} 2^V$

Understanding potentiometers

- How does the potentiometer work?
 - Outer terminals form a **resistor** ($10k\Omega$ in our case)
 - Connect the outer terminals to **+3.3V** and **ground**
- Inner terminal is the **wiper** which contacts an intermediate part of the resistor
 - i.e. there is one resistance from **+3.3V** to **wiper**; another from **wiper** to **ground**
- Two resistors form a **voltage divider**
 - **Ratio** of top and bottom resistors is what matters

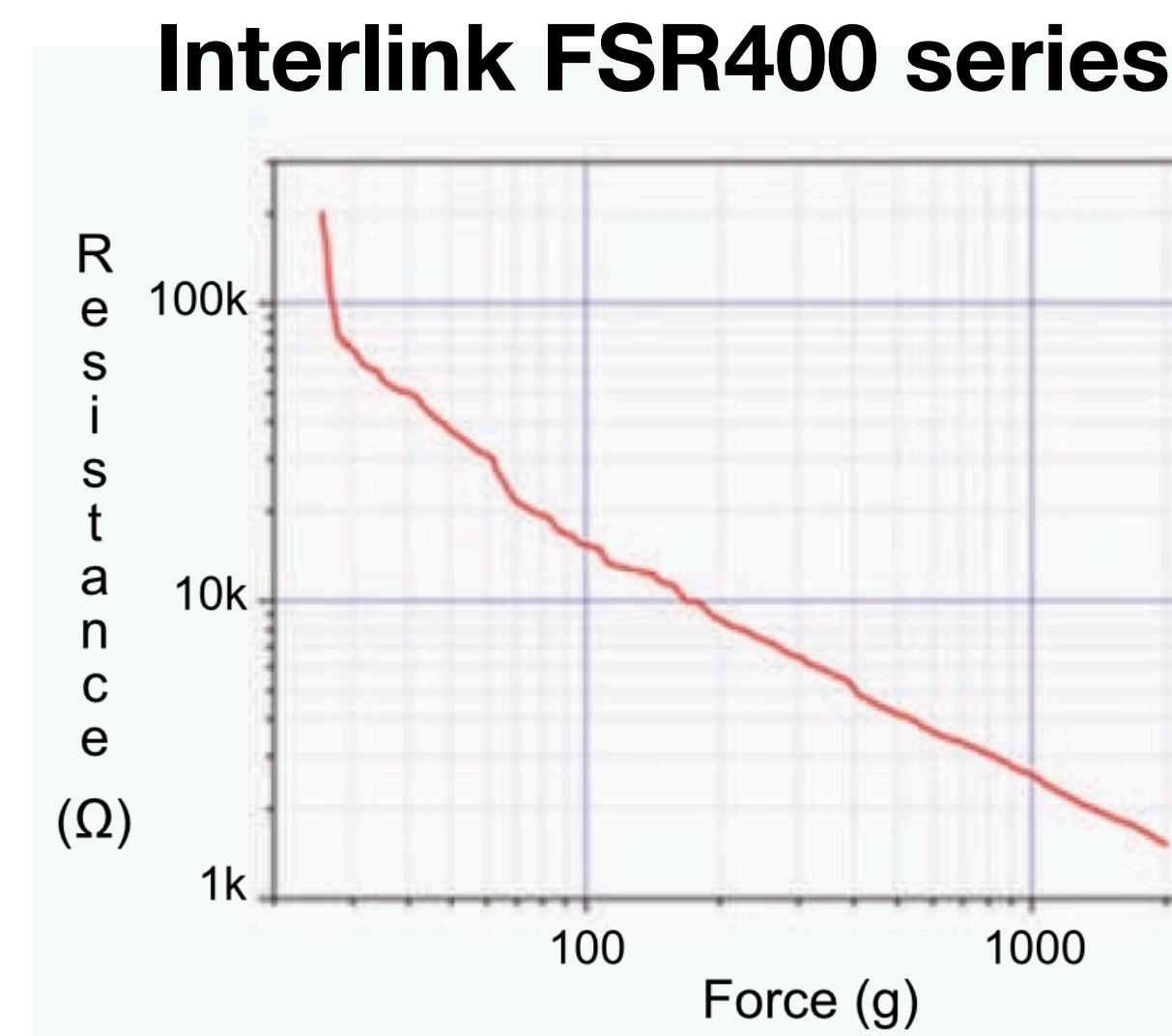


$$V_{out} = 3.3V \frac{R1B}{R1A + R1B}$$

$$R1A + R1B = 10k\Omega$$

Force-sensing resistors

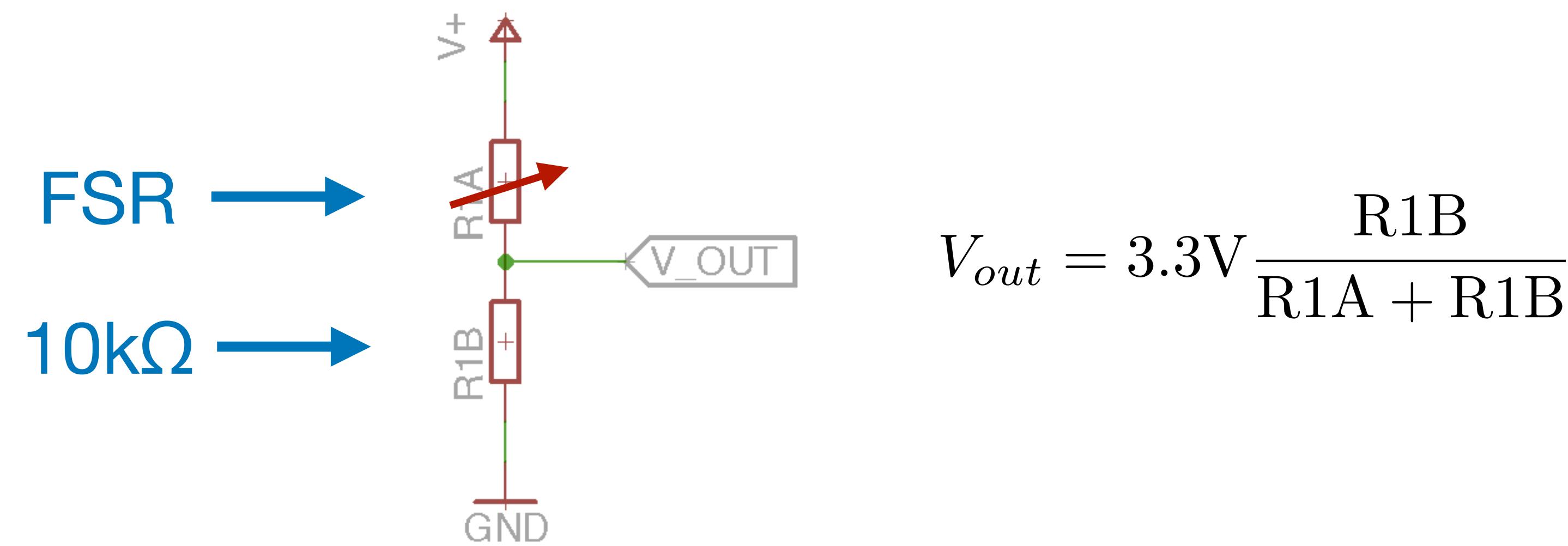
- A force-sensing resistor (FSR) measures pressure on its surface
 - They come in various shapes and sizes, but all work similarly
 - Increased pressure → decreased electrical resistance
 - Where would we find more details?
 - Read the datasheet!



Device Characteristics	
Actuation Force*	~0.2N min
Force Sensitivity Range*	~0.2N – 20N
Force Resolution	Continuous (analog)
Force Repeatability Single Part	+/- 2%
Force Repeatability Part to Part	+/- 6% (Single Batch)
Non-Actuated Resistance	>10 Mohms
Hysteresis	+10% Average (RF+ - RF-)/RF+

Force-sensing resistors

- How do we convert resistance into a **voltage** we can read?
 - Use a **voltage divider** circuit
 - One side is a fixed resistor (say 10k), the other side is the FSR

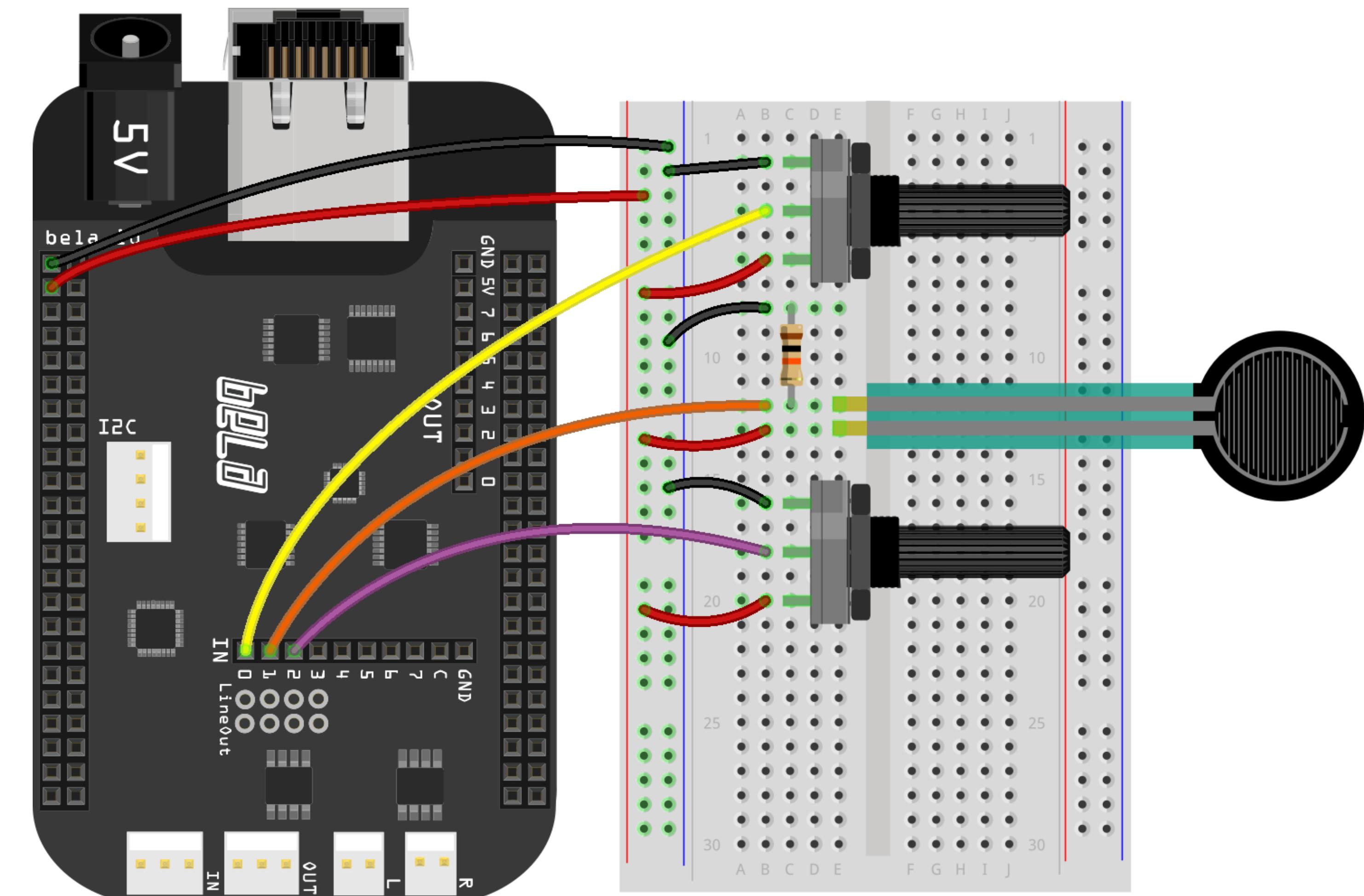


- The voltage at the midpoint moves toward the side with lower resistance
 - So in this case, the harder we press, the **higher** the voltage



A simple instrument

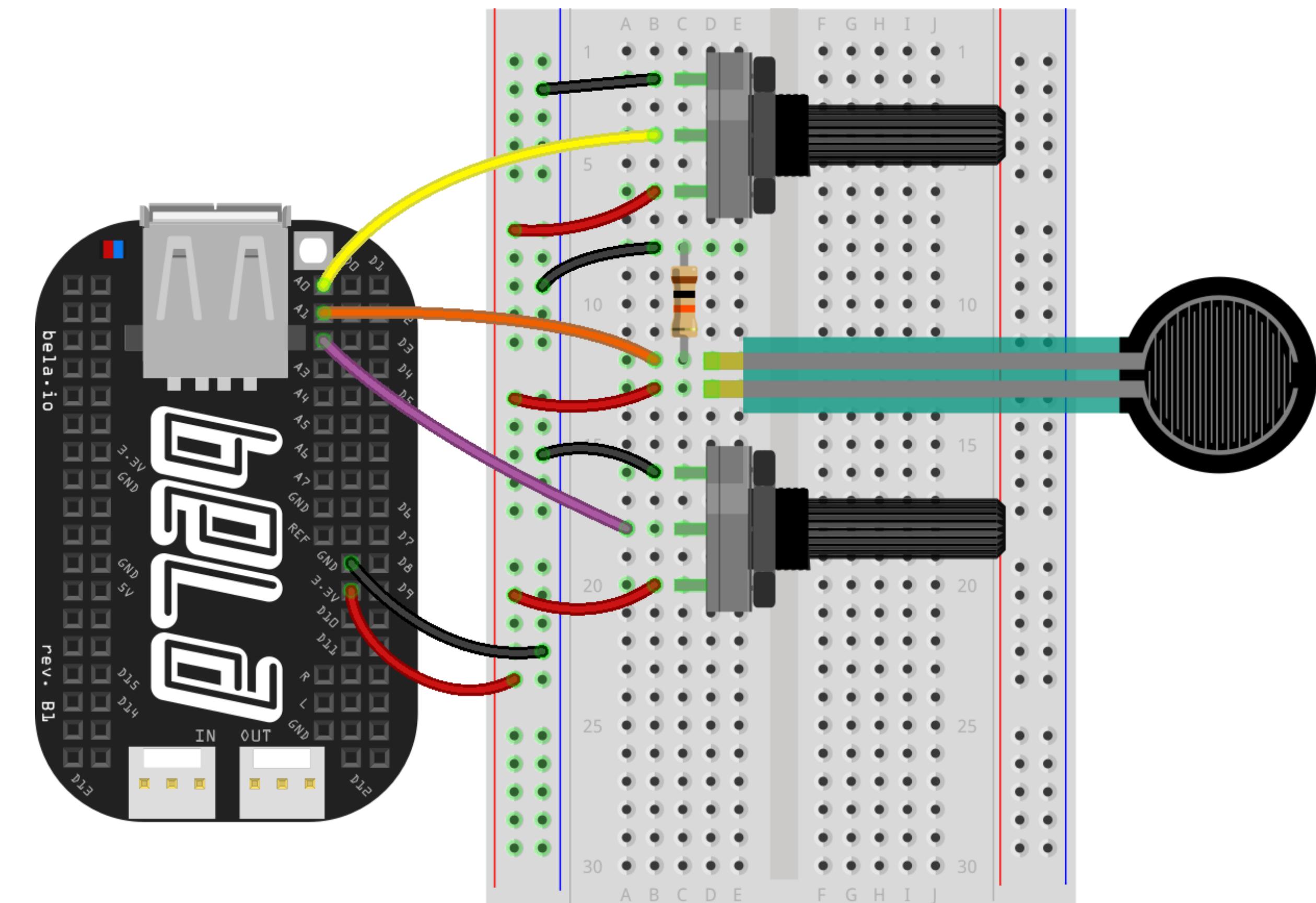
- Task: change your circuit so that the amplitude control uses an FSR rather than a potentiometer
 - ▶ Don't forget the 10k resistor!
 - ▶ Keep the code the same as before
 - ▶ How does the experience of playing it differ from before?
- Next, adjust the range for the FSR input to make a better playing experience



fritzing

A simple instrument

- Task: change your circuit so that the amplitude control uses an FSR rather than a potentiometer
 - ▶ Don't forget the 10k resistor!
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fritzing

Analog output

- Analog output works equivalently to analog input
 - Samples are generated at a **constant sample rate** (default 22.05kHz for 8 channels)
 - DC coupled (like analog input): can output **constant voltages**
 - Signal range of **0 to 1** corresponds to **0V to 5V**
 - Notice: this is different than analog input

- Main API:

```
void analogWrite(BelaContext *context, int frame, int channel, float value);
```

Reference to the **BelaContext** structure, which holds the actual data we want

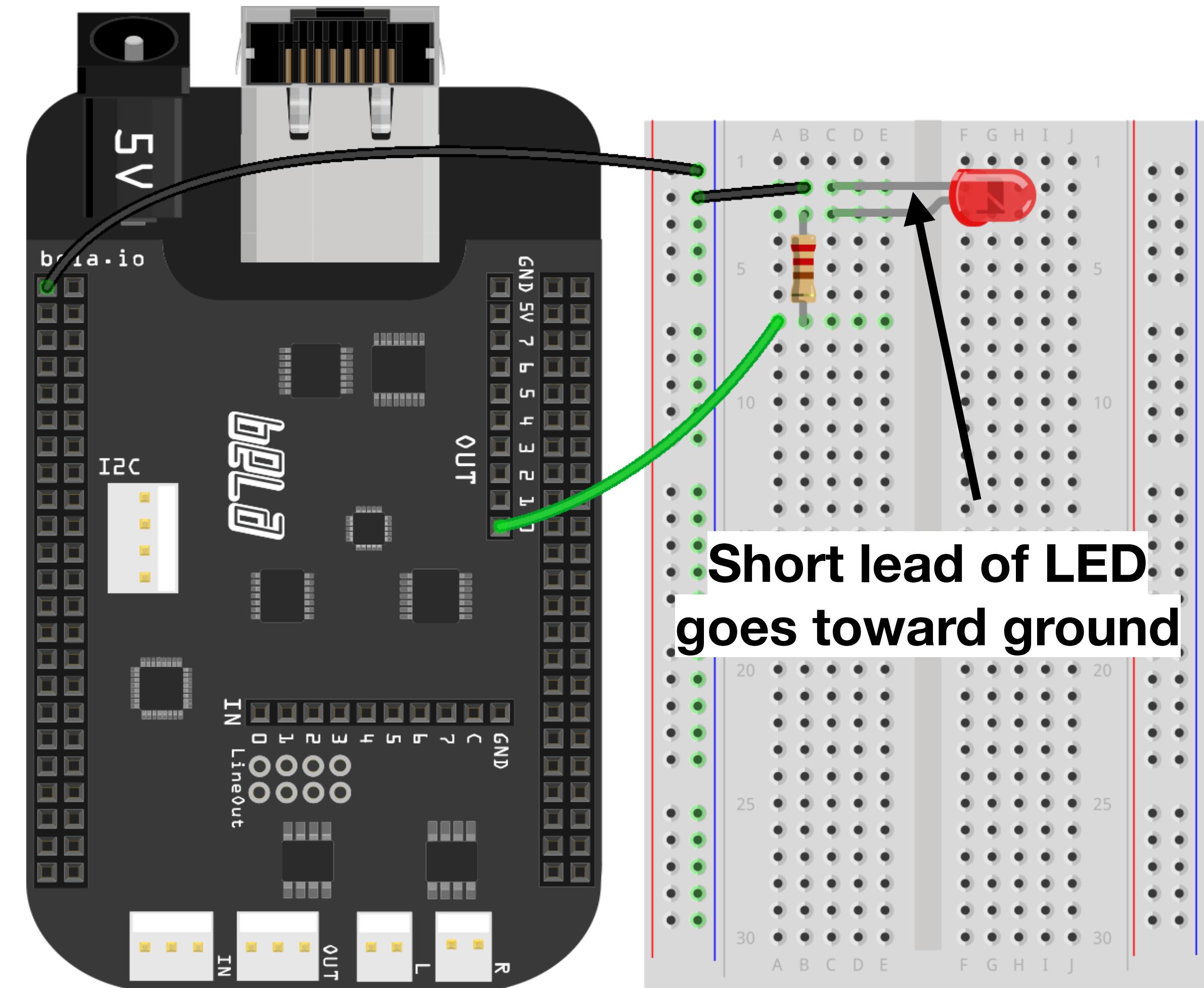
Which frame (i.e. **sample**) within the buffer to write

Which **channel** (e.g. 0, 1, ..., 7) to write

What **value** to write

Connect an LED (Bela only)

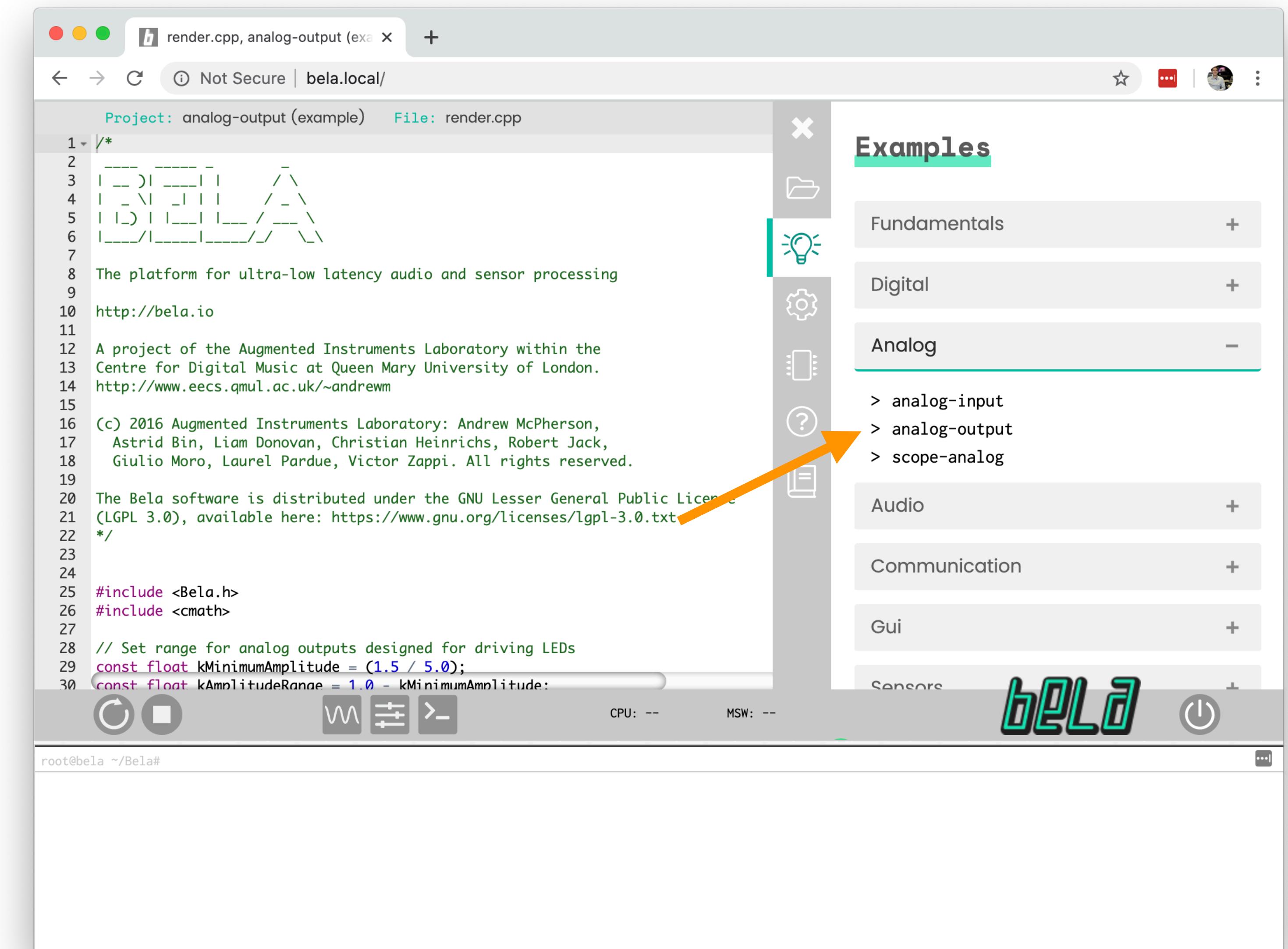
- Add this circuit to the breadboard (**don't remove the potentiometers or FSR**)
- It's important to use a **resistor** in series with your LED to limit the current that flows through it
- Open the **analog-output** example in the Bela IDE
- In this example, if you use multiple LEDs they will fade on and off with different **phase**



fritzing

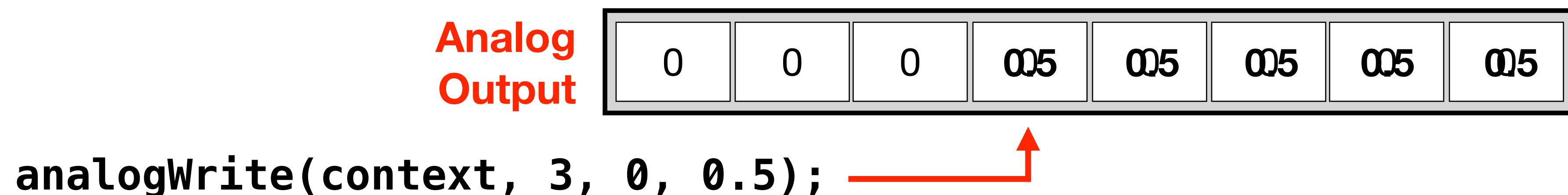
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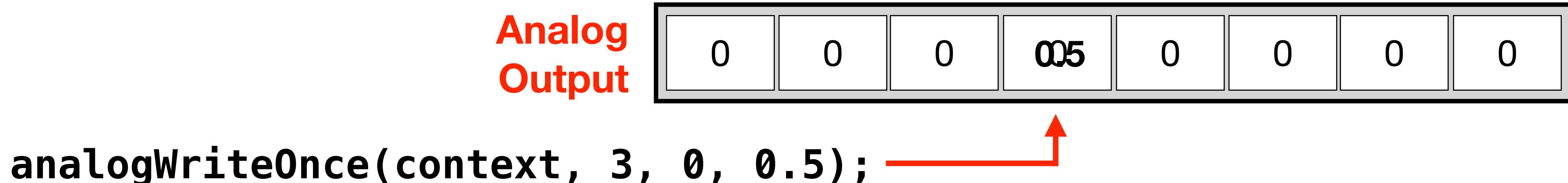


analogWrite() and analogWriteOnce()

- When writing an analog output, we might want the value to persist
 - Calling `analogWrite()` for a particular frame will also set the value for all future frames, until we call `analogWrite()` again
 - This is how writing a digital or analog pin on typical microcontrollers would work



- There is also a more efficient function that sets only the current frame
 - Use `analogWriteOnce()` if you will be writing a new value every frame anyway



VCO with LED

- The two detuned oscillators produce **beat frequencies**:

$$\sin(x) + \sin(y) = 2 \sin\left(\frac{x+y}{2}\right) \cos\left(\frac{x-y}{2}\right)$$

- Given that we have: $f_1 = (1+r)f$ $f_2 = (1-r)f$
- What is the beat frequency in terms of f and r ?
- **Task:** using the **vco-analog-out** example,
 - Add an **analog output** for the LED that pulses on and off at the **beat frequency**
 - Use **gLFO** to calculate the LED value, but at a lower frequency than the audio
 - Remember that the analog output has **half** the number of audio frames

Keep in touch!

Social media:

@BelaPlatform

forum.bela.io

blog.bela.io

More resources and contact info at:

learn.bela.io/resources