

# Real time digital audio processing with Arduino

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# Real time digital signal processing

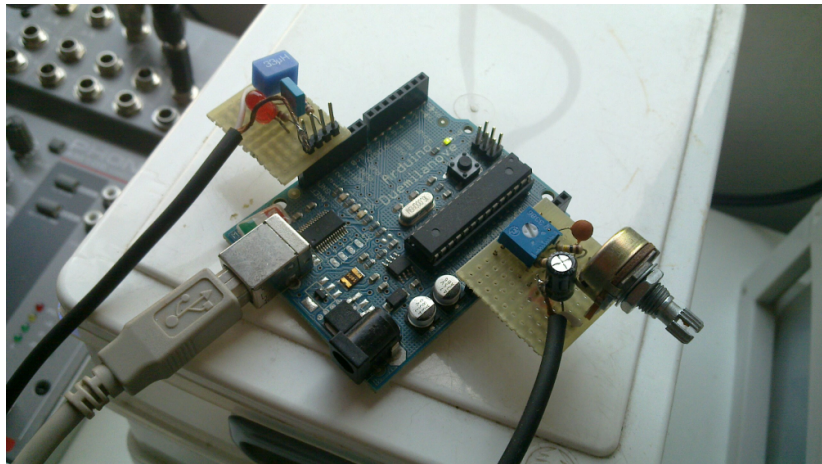
Digital audio signal processing includes:

- ▶ Acquiring samples.
- ▶ Processing.
- ▶ Outputting results.

Real time restriction:

- ▶ Block processing:  $N$  samples.
- ▶ Sampling frequency:  $R$  Hz.
- ▶ DSP cycle period:  $T_{DSP} = \frac{N}{R}$  s.

# Real time DSP with Arduino



<http://interface.khm.de/index.php/lab/experiments/arduino-realtime-audio-processing/>

# Atmel AVR microcontroller (ATmega328P)

Microcontroller's characteristics:

- ▶ CPU: ALU and registers (16 MHz - 8 bits).
- ▶ Memory: Flash (32 KB), SRAM (2 KB) e EEPROM (1 KB).
- ▶ Digital I/O ports:
  - ▶ Audio input: analog to digital converter.
  - ▶ Audio output: counters capable of doing PWM.

# Arduino performance for real time digital audio processing

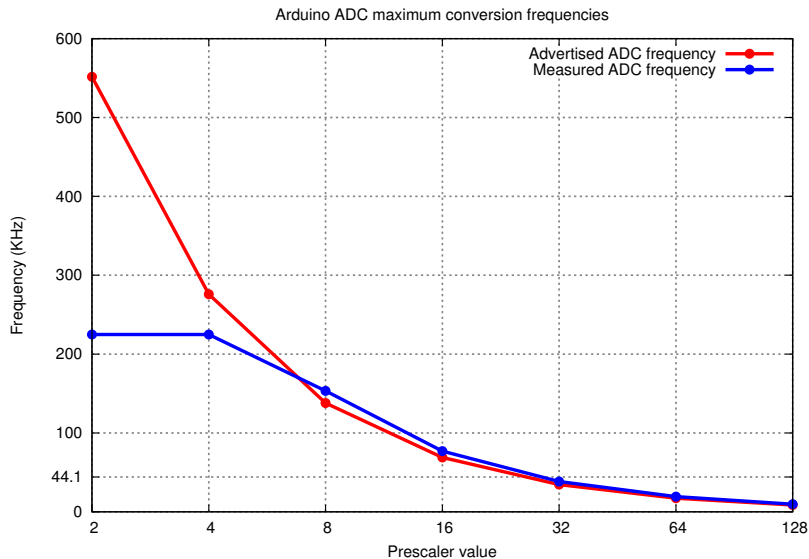
## Questions:

- ▶ What is the maximum number of operations feasible in real-time?
- ▶ Which implementation details make a difference?
- ▶ What is the quality of the resulting audio signal?

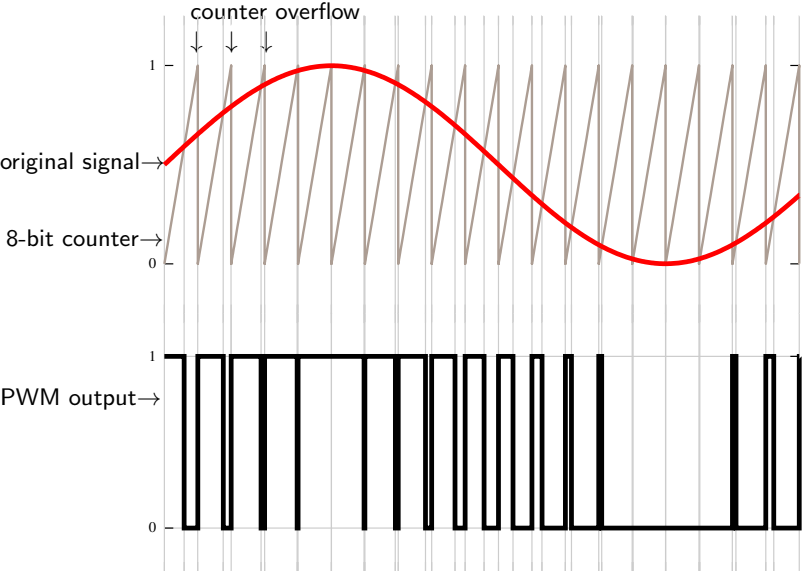
## DSP algorithms implemented:

- ▶ Additive synthesis.
- ▶ Time-domain convolution.
- ▶ FFT.

# Audio input: analog to digital converter



# Pulse Width Modulation



# Audio output: Pulse Width Modulation

8-bit counter frequencies for different prescaler values:

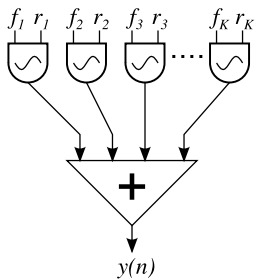
| prescaler | $f_{incr}$ (KHz) | $f_{overflow}$ (Hz) |
|-----------|------------------|---------------------|
| 1         | 16000            | 62500               |
| 8         | 2000             | 7812                |
| 32        | 500              | 1953                |
| 64        | 250              | 976                 |
| 128       | 125              | 488                 |
| 256       | 62.5             | 244                 |
| 1024      | 15.625           | 61                  |

PWM overflow interrupt allow for periodically triggering:

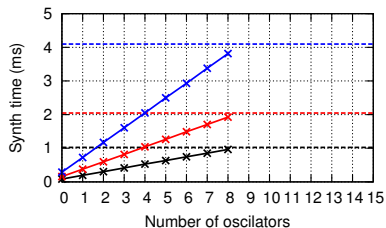
- ▶ ADC conversion.
- ▶ Signal manipulation.
- ▶ PWM mechanism value set.



# Additive synthesis

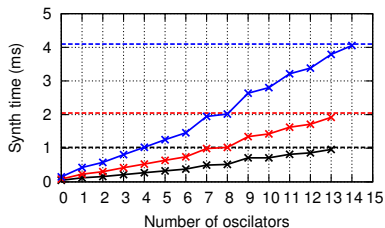


Additive Synthesis on Arduino (loop)



bl. size 32 —\*—  
bl. size 64 —\*—  
bl. size 128 —\*—

Additive Synthesis on Arduino (inline)



bl. size 32 —\*—  
bl. size 64 —\*—  
bl. size 128 —\*—

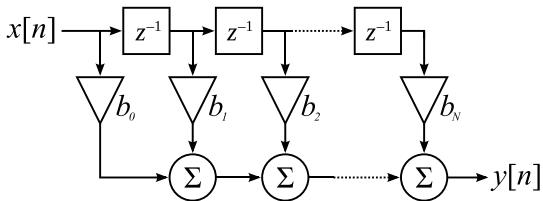
# Additive synthesis

## Example

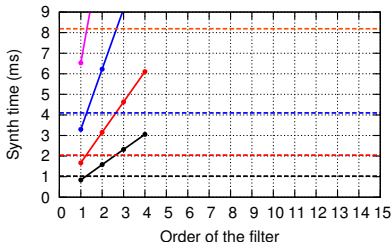
Sum of harmonics with  $f_0=200$  Hz:

$$y[n] = \sum_{k=1}^? \sin \left( 2\pi k 200 \frac{n}{R} \right).$$

# Time-domain convolution

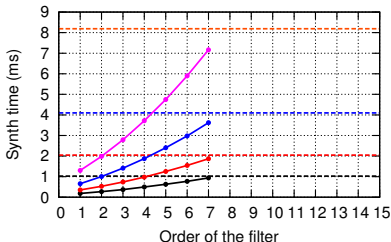


Time-domain convolution on Arduino (mult/div)



bl. size 32 —●—      bl. size 128 —●—  
 rt per. 32 - - - - -      rt per. 128 - - - - -  
 bl. size 64 —●—      bl. size 256 —●—  
 rt per. 64 - - - - -      rt per. 256 - - - - -

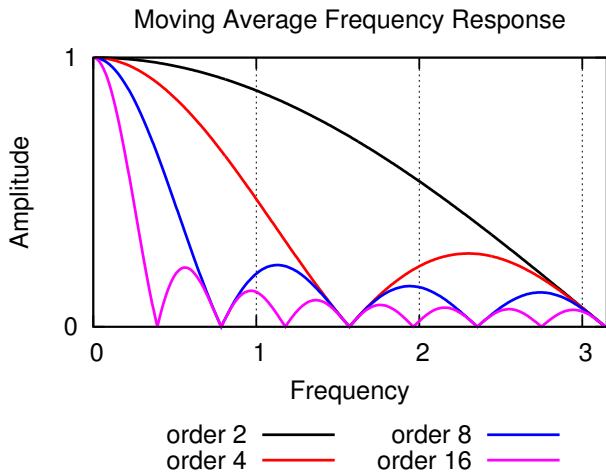
Time-domain convolution on Arduino (bit-shifting)



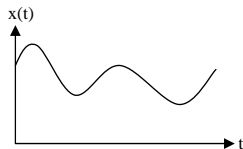
bl. size 32 —●—      bl. size 128 —●—  
 rt per. 32 - - - - -      rt per. 128 - - - - -  
 bl. size 64 —●—      bl. size 256 —●—  
 rt per. 64 - - - - -      rt per. 256 - - - - -

# Time-domain convolution

Example: moving average

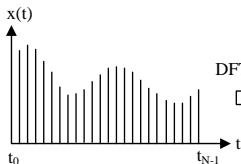


# Fast Fourier Transform



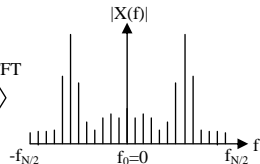
1) continuous signal in time domain

ADC  
⇒

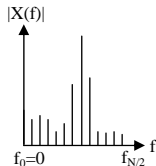


2)  $N$  points in time domain

DFT/FFT  
⇒



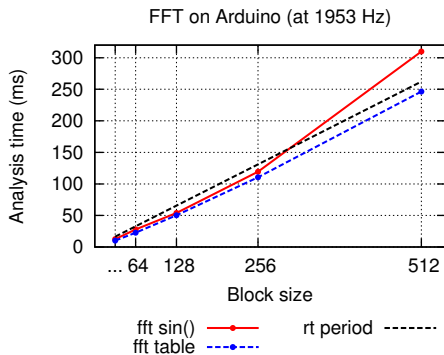
3)  $N$  points in frequency domain containing both negative and positive frequency parts



4)  $N/2+1$  points in amplitude/power spectrum



# Fast Fourier Transform



Maximum frequency for block size 256:

- ▶ Mean calculation time  $\approx 428,15 \mu\text{s}$  per sample.
- ▶ Maximum frequency  $\approx 2.335 \text{ Hz}$ .
- ▶ PWM prescaler value 32  $\Rightarrow R = 1.953 \text{ Hz}$ .

# Conclusions

- ▶ Many implementation details make a difference:
  - ▶ Types used (byte, unsigned long, int, float, etc).
  - ▶ Type of operations: integer (multiplication, division, sum) and bitwise.
  - ▶ Presence of loops.
  - ▶ Use of variables and vectors.
  
- ▶ Families of algorithms can be found to make it feasible to use the Arduino in real time audio processing.

# Thank you for your attention!

## Contact:

- ▶ Email: {ajb,mqz}@ime.usp.br
- ▶ This presentation: <http://www.ime.usp.br/~ajb/>
- ▶ CM at IME: <http://compmus.ime.usp.br/>

## Attribution of figures taken from wikipedia:

- ▶ PWM: Zurecs (zureks@gmail.com).
- ▶ Additive synthesis: Chrisjonson.
- ▶ FFT: Virens.