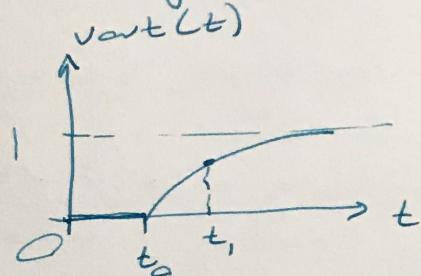
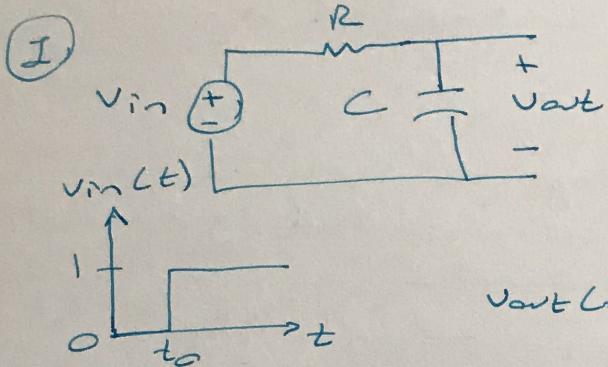


Conversion Between Analog and Digital Values

- Analog and Digital Values
 - Analog Values in Physical systems
 - Digital Values in Embedded systems
 - Digital Values in code
- Analog to Digital Conversion (ADC)
 - Sampling
 - Quantization
 - ADC under Mbed
- Digital to Analog Conversion (DAC)
 - Zero-Order Hold
 - Pulse Width Modulation
 - DAC under Mbed
- Mbed Example Codes

Analog Values in Physical Systems



$v_{out}(t)$

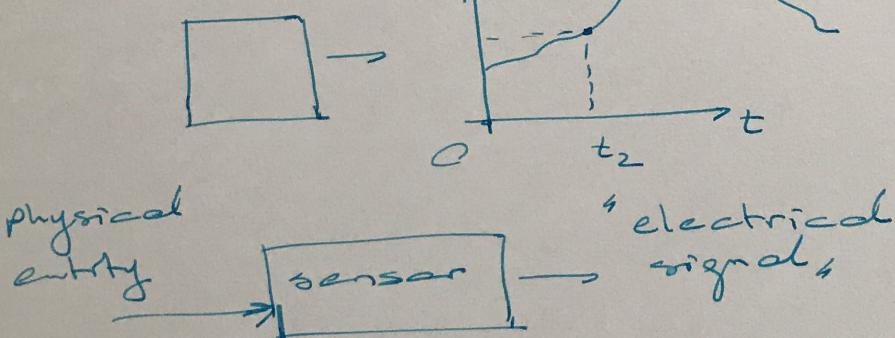
① defined for all time values.
② amplitude at any specific time is "real" valued.

- ① → We have infinite number of values
- ② → The amplitude at any time can take infinite number of values.

In our microcontroller, we do not have infinite storage space.

It is also true for any embedded system.

(II) Temperature sensor



- ①
②

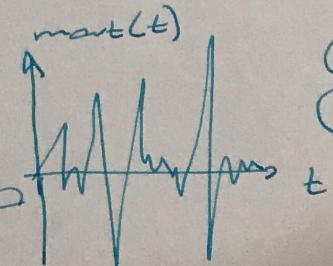
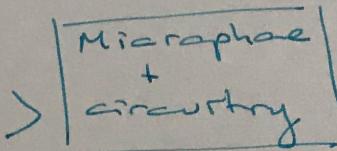
also hold here.

(III) Microphone

Audio signals

"Is a speaking now"

<))>



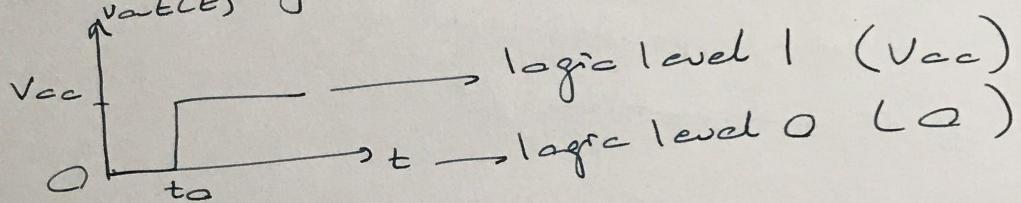
- ①
②

also hold here.

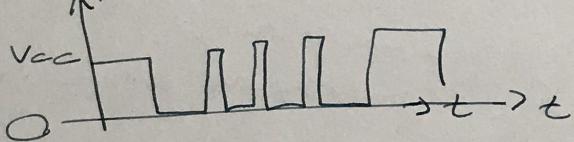
How can I represent of process analog values in my microcontroller?

- Microcontroller has limited memory.
- I need "another" representation for this purpose.
- ⇒ Digital value.
- The simplest digital form is a "bit" which can take two values as logic level 0 or 1.

RC-circuitry



Audio signal

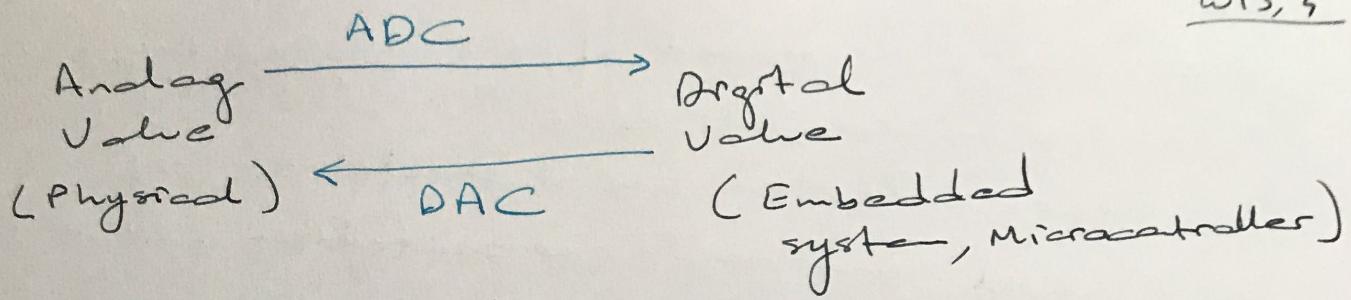


→ We use "group" of bits, formatted to represent a value in embedded system.

→ In C code, "formatted values"

`char next;` → 8-bits, $(-128, 127)$ integer.
`int next;` → 32-bits, $(2^{-31}, 2^{31}-1)$ integer.
`float next;` → 32-bits; ⇒ floating point representation
 integer, fractional parts.

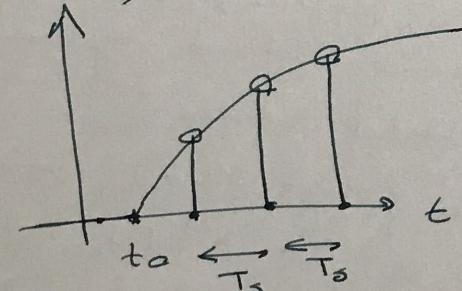
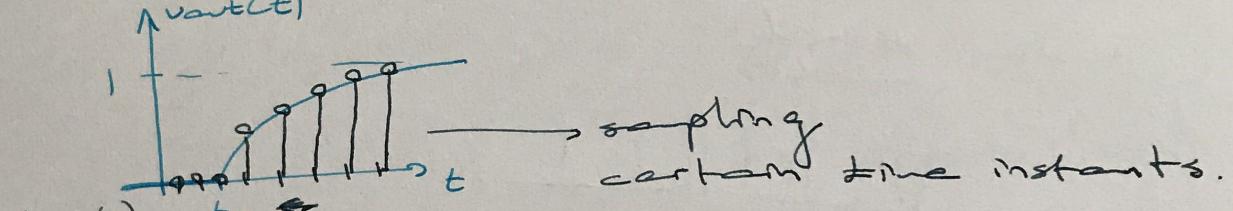
⇒ Digital form



Analog to Digital Conversion

- ① We have to limit the number of samples in our analog signal.
⇒ Sampling (*)
- ② We also have to limit the analog value by using "predefined" number of bits.
⇒ Quantization (*)

- ① RC circuit



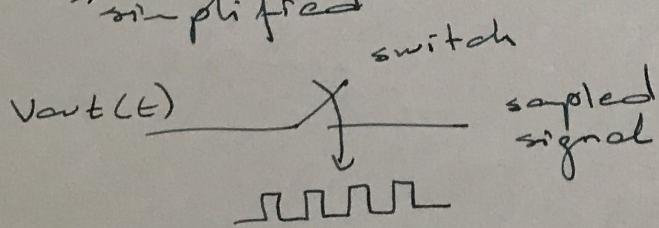
I sample the analog signal periodically.

$T_s \rightarrow$ sampling period

$$f_s = \frac{1}{T_s} \rightarrow$$
 sampling frequency

Sampling operation

"simplified"



Index values to our sampled signal.

1. sample
2. sample
3. sample

⋮

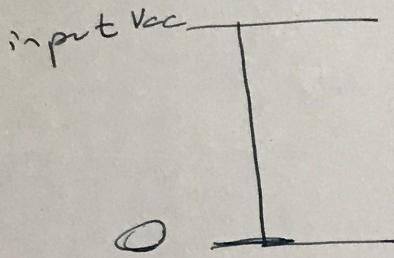
$\times [n]$

↳ index $\in I$

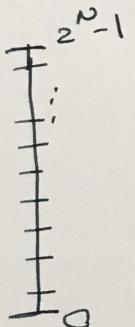
"Array" in C language.

I have "sampled" the values, however they still have infinite precision.

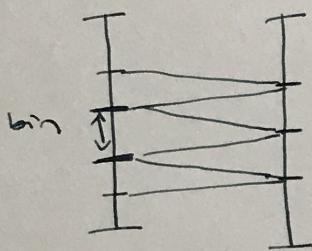
Quantization



→ Quantization



N-bit quantization.

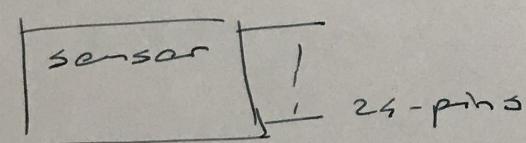


a range in input analog value
is represented by a single
number.

"12-bit" ADC module \Rightarrow 0, V_{ac} volts input range is represented by digital values
 $0 \rightarrow 2^{12}-1 \Rightarrow$ in C language
unsigned int.

Sensor 24-bits

\Rightarrow It has its own "ADC" module.



or it uses digital com.
module for operation
UART, SPI, I²C ---

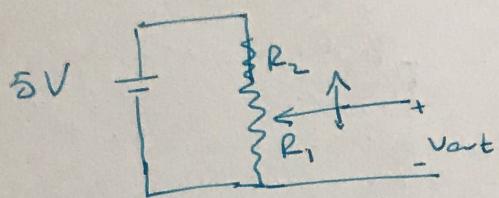
analog \rightarrow $\boxed{\text{ADC}}$ \rightarrow digital

1. sampling
2. quantization

\rightarrow ADC under Microcontroller

\Rightarrow my microcontroller "must" have the ADC module within it.

Potentiometer (Analog voltage source)



$$V_{out} = \frac{5 R_1}{R_1 + R_2}$$

$$V_{out} \in 0 - 5V$$

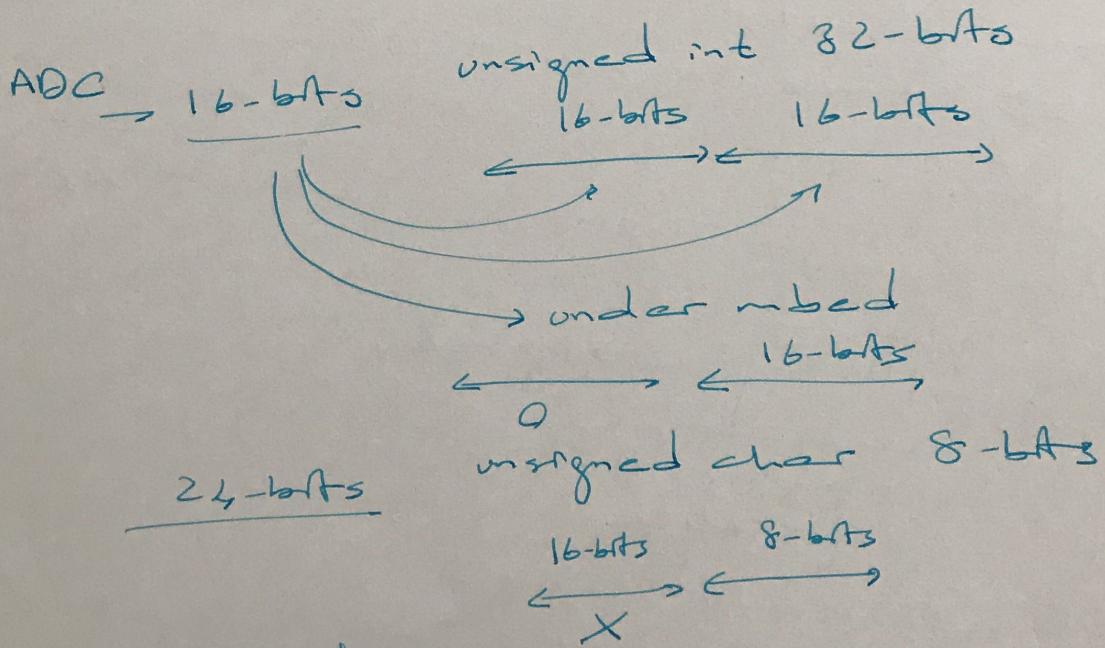
Mbed "adc-exapcl.cpp"

AnalogIn ain(p15);

ain.read()

5 I → "float output, 32-bit
0 - 1"

ain.read() → 0x0400
5 I → unsigned integer "16-bits"
0 → 0x0000



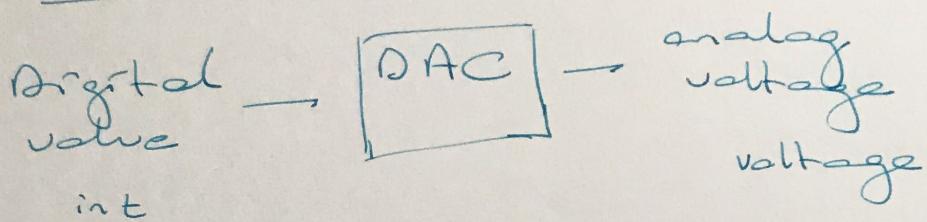
⇒ Quantization.

wait_ms(200)

→ sampling at every 200 msec.

Digital to Analog Converter

w13, 7



- The microcontroller has a specific
 ① Digital to Analog Converter (DAC) module.

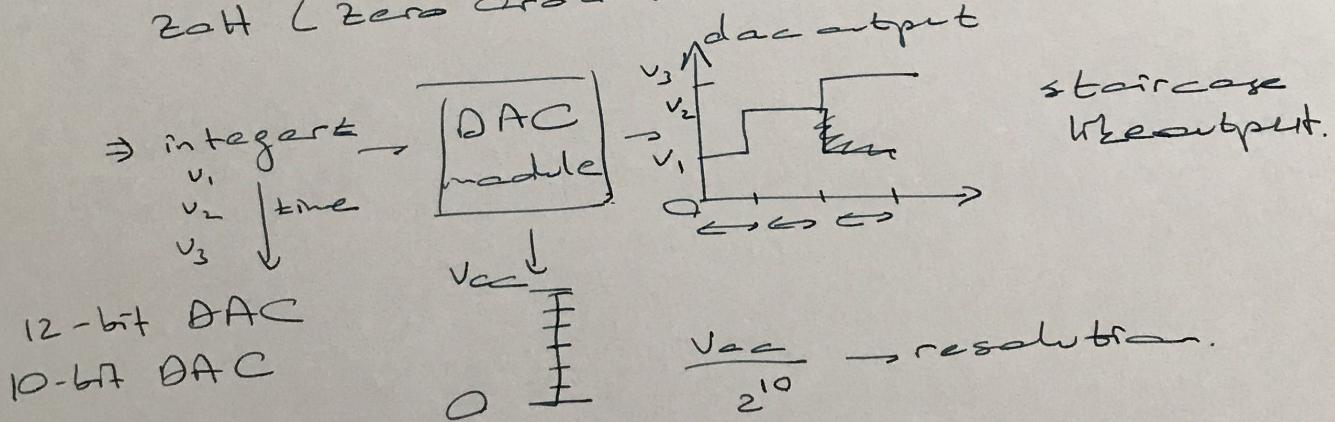
⇒ This is not mandatory.

If we have the DAC module, 2¹⁰ circuit.

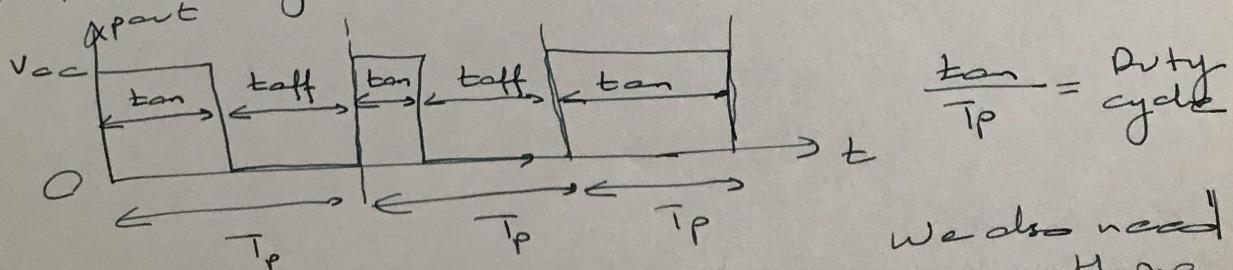
- Pulse Width Modulation (PWM)
 ② PWM signal can be generated by timers.
 ⇒ This option is more popular.

- ① I have the DAC module.

2¹⁰ (Zero Order Hold) setup.



- ② PWM signal



Average output voltage

$$V_{avg} = \frac{t_{on}}{T_p} V_{cc}$$

→ We also need a smoothing filter.