

# Microprocessors & Interfacing

## *Analog Input/Output*

Lecturer : Dr. Annie Guo

## Lecture overview

- Analog output
  - PMW
  - Digital-to-Analog (D/A) Conversion
- Analog input
  - Analog-to-Digital (A/D) Conversion

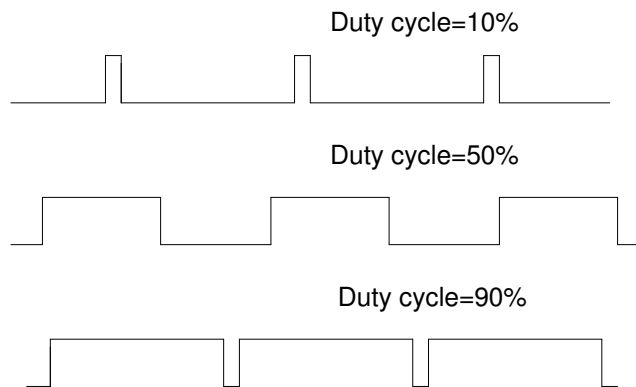
## PWM Analog Output

- PWM (Pulse Width Modulation) is a way of digitally encoding analog signal levels.
  - Through the use of high-resolution counters, the duty cycle (pulse width/period) of a pulse wave is modulated to encode a specific analog signal level.
- The PWM signal is still digital
  - Its value is either full high or full low.
  - Given a sufficient bandwidth, any analog value can be encoded with PWM.
- PWM is a powerful technique for controlling analog circuits with a processor's digital outputs.
- It is employed in a wide variety of applications
  - E.g. motor speed control

## PWM Analog Output (cont.)

- A low-pass filter is required to smooth the input signal and eliminate the inherent noise components in PWM signal.
- The output voltage is directly proportional to the pulse width.
  - By changing the pulse width of the PWM waveform, we can control the output value.

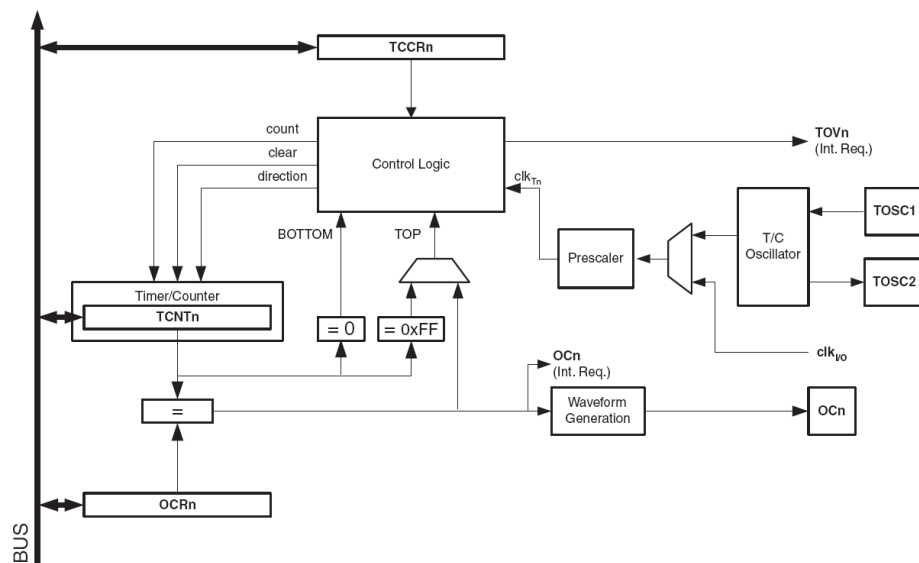
## Examples of PWM Signals



## PWM Generation In AVR

- PWM can be obtained through the provided timers.

## Recall: Timer0



## Configuration for PWM

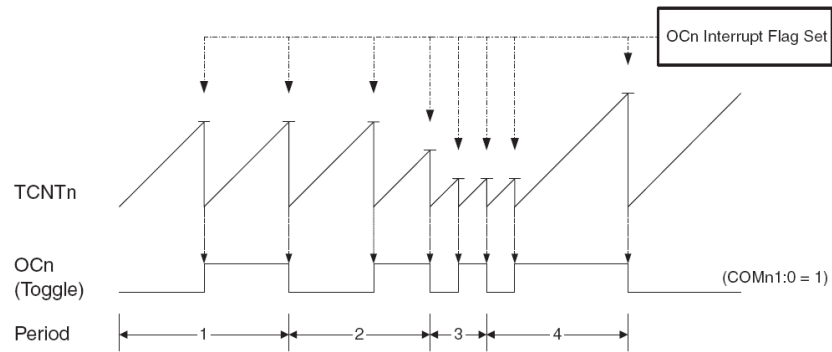
- TCCR0

7	6	5	4	3	2	1	0
FOC0	WGM00	COM01	COM00	WGM01	CS02	CS01	CS00

Mode	WGM01 (CTC0)	WGM00 (PWM0)	Timer/Counter Mode of Operation	TOP	Update of OCR0 at	TOV0 Flag Set on
0	0	0	Normal	0xFF	Immediate	MAX
1	0	1	PWM, Phase Correct	0xFF	TOP	BOTTOM
2	1	0	CTC	OCR0	Immediate	MAX
3	1	1	Fast PWM	0xFF	TOP	MAX

## CTC

- Clear Timer on Compare Match

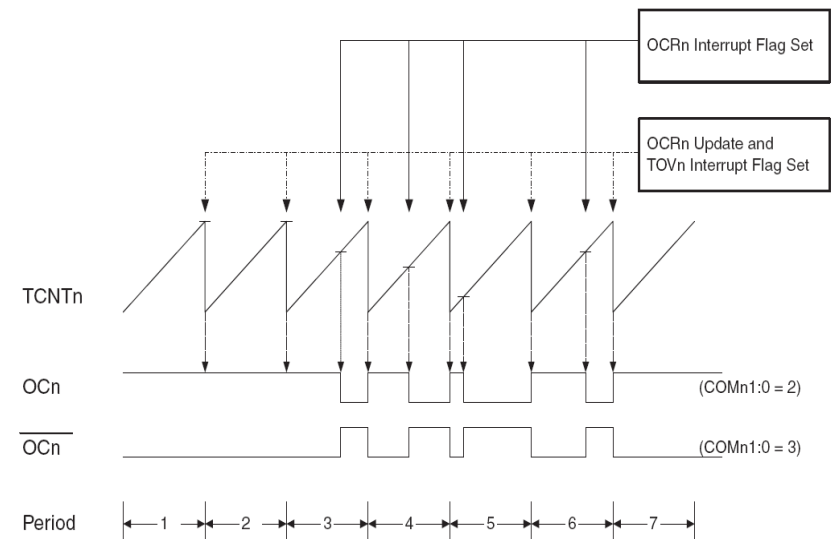


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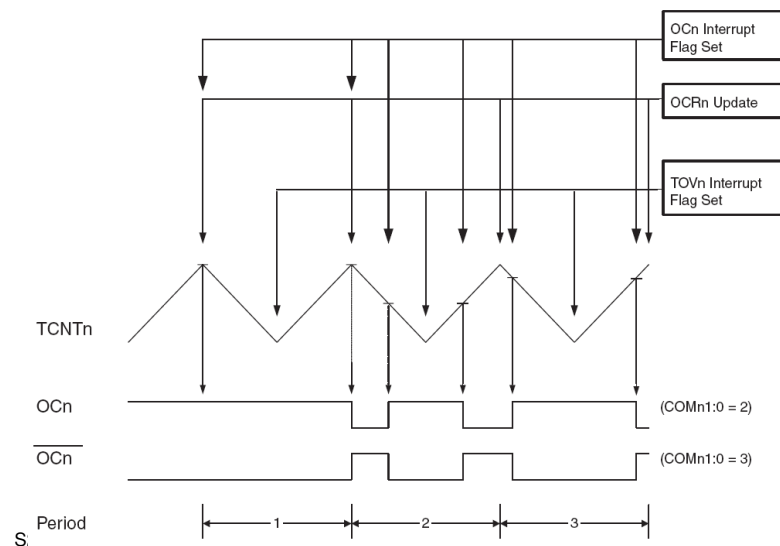
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## Fast PWM



## Phase Correct PWM



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## Example

- Generate a PWM waveform.

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## Example (solution)

- Use Timer2
  - Set OC2 as output
  - Set the Timer2 operation mode as Phase Correct PWM mode
  - Set the timer clock

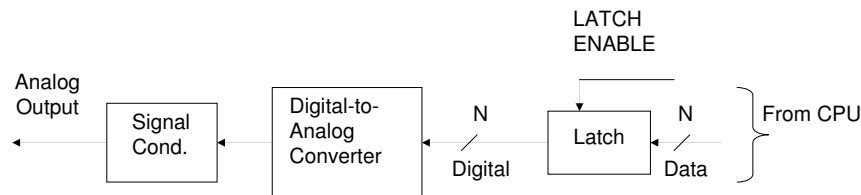
## Example Code

```
.include "m64def.inc"
.def temp=r16

    ldi temp, 0b10000000
    out DDRB, temp           ; Bit 7 will function as OC2.

    ldi temp, 0x4A           ; the value controls the PWM duty cycle
    out OCR2, temp
                             ; Set the Timer2 to Phase Correct PWM mode.
    ldi temp, (1<<WGM20)|(1<<COM21)|(1<<CS20)
    out TCCR2, temp
```

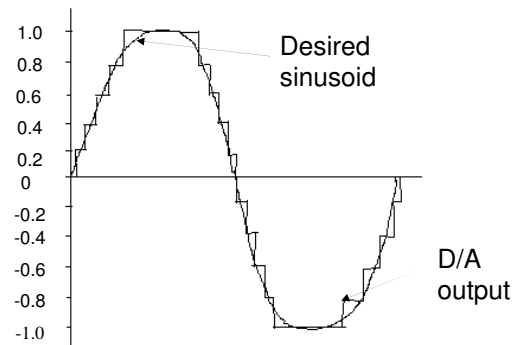
## Digital-to-Analog Conversion



## Digital-to-Analog Conversion (cont.)

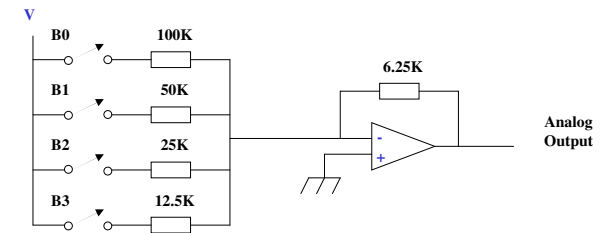
- A parallel output interface connects the D/A to the CPU.
- The latches may be part of the D/A converter or the output interface.
- Digital value is converted into continuous value.
- A signal conditioning block may be used as a filter to smooth the quantized nature of the output.
  - The signal conditioning block also provide isolation, buffering and voltage amplification if needed.

## Quantized D/A Output



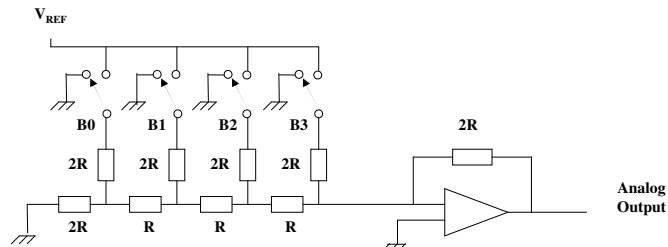
## Binary-weighted D/A Converter

- As the switches for the bits are closed, a weighted current is supplied to the summing junction of the amplifier.
- For high-resolution D/A converters, the binary-weighted type must have a wide range of resistors. This may lead to temperature stability and switching problems.



## R-2R Ladder D/A Converter

- As a switch changes from the grounded position to the reference position, a binary-weighted current is supplied to the summing junction.
- For high-resolution D/A converters, a wide range of resistors are not required.



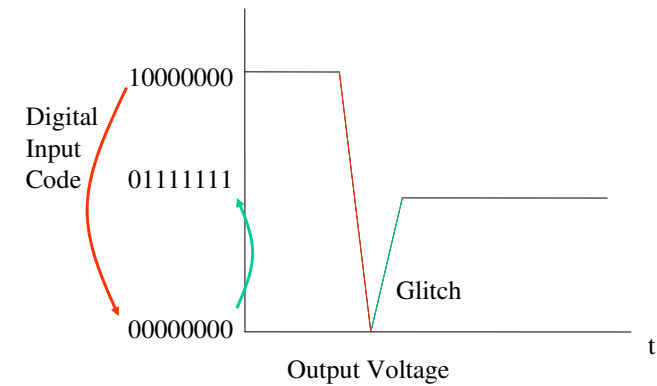
## D/A Converter Specifications

- Resolution and linearity.
  - The resolution is determined by the number of bits and is given as the output voltage corresponding to the smallest digital step, i.e. 1 LSB.
  - The linearity shows how closely the output voltage to the idea values (a straight line drawn through zero and full-scale).
- Settling Time.
  - The time taken for the output voltage to settle to within a specified error band, usually  $\pm \frac{1}{2}$  LSB.

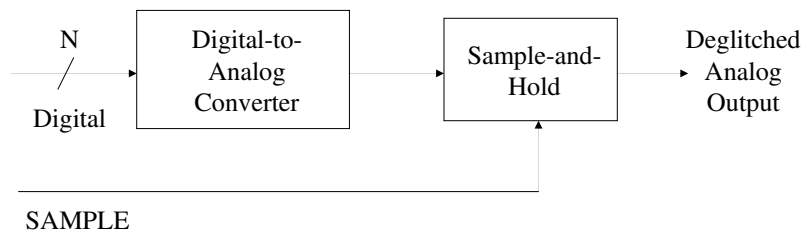
## D/A Converter Specifications (cont.)

- Glitches.
  - A glitch is caused by asymmetrical switching in the D/A switches. If a switch changes from a one to a zero faster than from a zero to a one, a glitch may occur.
    - Consider changing the output code of a 8-bit D/A from 10000000 to 01111111 in the next slide.
  - D/A converter glitch can be eliminated by using a sample-and-hold.

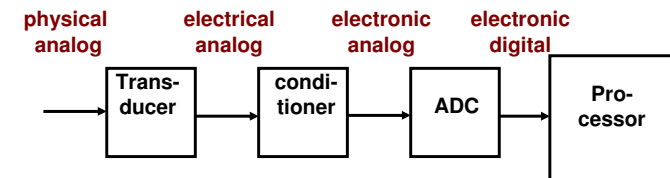
## D/A Output Glitch



## Deglitched D/A



## A/D Conversion



## Data Acquisition and Conversion

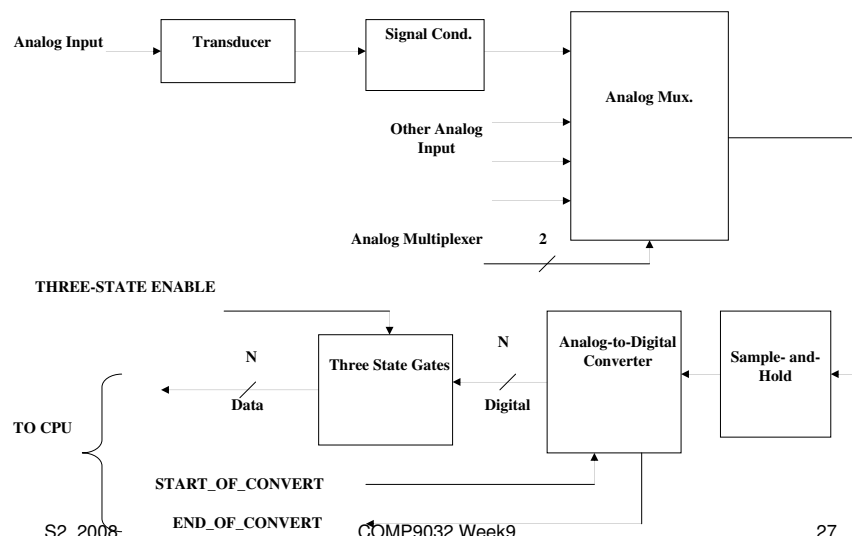
Procedure of data acquisition and conversion:

- A transducer converts physical values to electrical signals, either voltages or currents.
- Signal conditioner performs the following tasks:
  - Isolation and buffering: The input to the A/D may need to be protected from dangerous voltages such as static charges or reversed polarity voltages.
  - Amplification: Rarely does the transducer produce the voltage or current needed by the A/D. The amplifier is designed so that the full-scale signal from the analog results in a full-scale signal to the A/D.
  - Bandwidth limiting: The signal conditioning provides a low-pass filter to limit the range of frequencies that can be digitized.

## Data Acquisition and Conversion (Cont.)

- In applications where several analog inputs must be digitized, an analog multiplexer is followed the signal conditioning. It allows multiple analog inputs, each with its own signal conditioning for different transducers.
- The sample-and-hold circuit samples the signal and holds it steady while the A/D converts it.
- The A/D converter converts the sampled signal to digital values.
- The three state gates hold the digital values generated by the A/D converter.

## Data Acquisition System



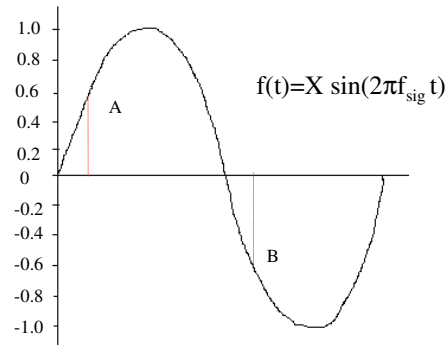
## Shannon's Sampling Theorem

Claude Shannon's Theorem:

- When a signal,  $f(t) = X \sin(2\pi f_{\text{sig}} t)$ , is to be sampled (digitized), the minimum sampling frequency must be twice the signal frequency.

## Sample Examples

- Sampled at twice of the signal frequency.



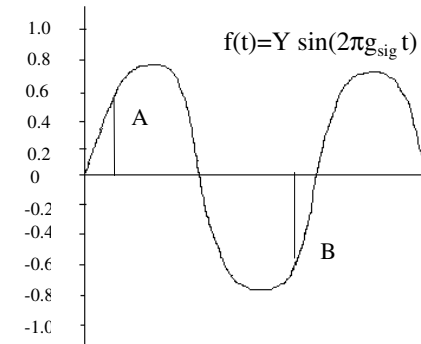
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## Sample Examples

- Under-sampled, with sample frequency less than twice of the signal frequency



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## Shannon's Sampling Theorem and Aliasing

- To preserve the full information in the signal, it is necessary to sample at twice the maximum frequency of the signal. This is known as the **Nyquist rate**.
- A signal can be exactly reproduced if it is sampled at a frequency  $F$ , where  $F$  is greater than or equal to the Nyquist rate.
- If the sampling frequency is less than Nyquist rate, the waveform is said to be under-sampled.

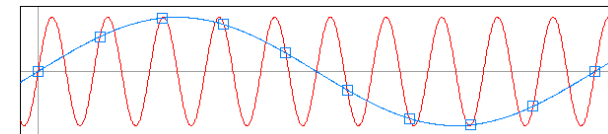
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## Shannon's Sampling Theorem and Aliasing (Cont.)

- Undersampled signal, when converted back into a continuous time signal, will exhibit a phenomenon called *aliasing*.
  - Aliasing is the presence of unwanted components in the reconstructed signal. These components were not present when the original signal was sampled.



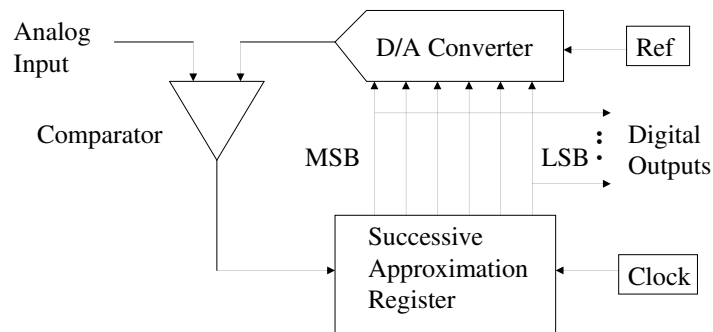
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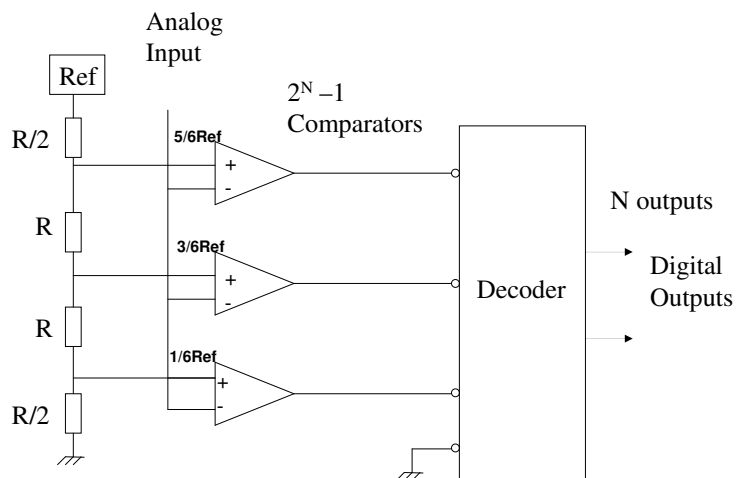
## Successive Approximation Converter



## Successive Approximation A/D Converter

- Each bit in the successive approximation register is tested, starting at the most significant bit and working toward the least significant bit.
- As each bit is set, the output of the D/A converter is compared with the input.
- If the D/A output is lower than the input signal, the bit remains set and the next bit is tried.
- N times are required to set and test each bit in the successive approximation register.

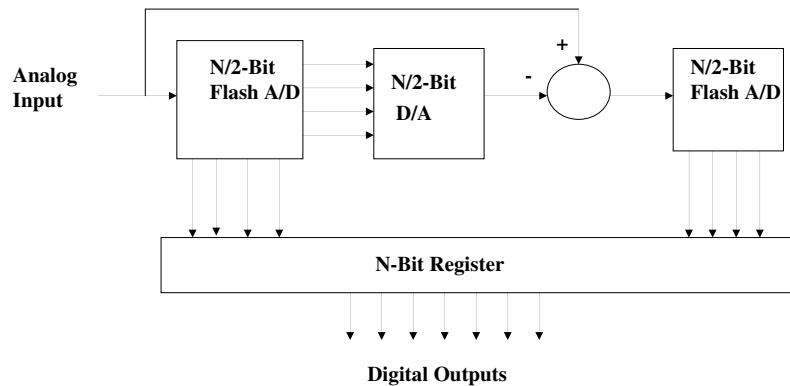
## Parallel A/D Converter



## Parallel A/D Converter

- An array of  $2^N - 1$  comparators and produces an output code in the propagation time of the comparators and the output decoder.
- Fast but more costly in comparison to other designs.
- Also called flash A/D converter.

## Two-Stage Parallel A/D Converter



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## Two-Stage Parallel A/D Converter

- The input signal is converted in two pieces.
  - First, a coarse estimate is found by the first parallel A/D converter. This digital value is sent to the D/A and summer, where it is subtracted from original signal.
  - The difference is converted by the second parallel converter and the result combined with the first A/D to give the digitized value.
- It has nearly the performance of the parallel converter but without the complexity of  $2^N - 1$  comparators.
- It offers high resolution and high-speed conversion for applications like video signal processing.

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## A/D Converter Specifications

- Conversion time
  - The time required to complete a conversion of the input signal.
  - Establishes the upper signal frequency limit that can be sampled without aliasing.
$$f_{MAX} = 1 / (2 * \text{conversion time}) \quad (1)$$
- Resolution
  - The number of bits in the converter gives the resolution and thus the smallest analog input signal for which the converter will produce a digital code.
  - It may be given in terms of the full-scale input signal:
 
$$\text{Resolution} = \text{full-scale signal} / 2^n \quad (2)$$
  - It is often given as the number of bits,  $n$ ; or stated as one part in  $2^n$ .
  - Sometimes it is given as a percent of maximum.

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## A/D Converter Specifications (Cont.)

- Accuracy
  - Relates to the smallest signal (or noise) to the measured signal.
  - Given as a percent and describes how close the measurement is to the actual value.
$$\text{The signal is accurate to within } 100\% * V_{RESOLUTION} / V_{SIGNAL} \quad (3)$$
- Linearity
  - The derivation in output codes from the real value (a straight line drawn through zero and full-scale).
  - The best that can be achieved is  $\pm 1/2$  of the least significant bit
 
$$(\pm 1/2 \text{ LSB}).$$

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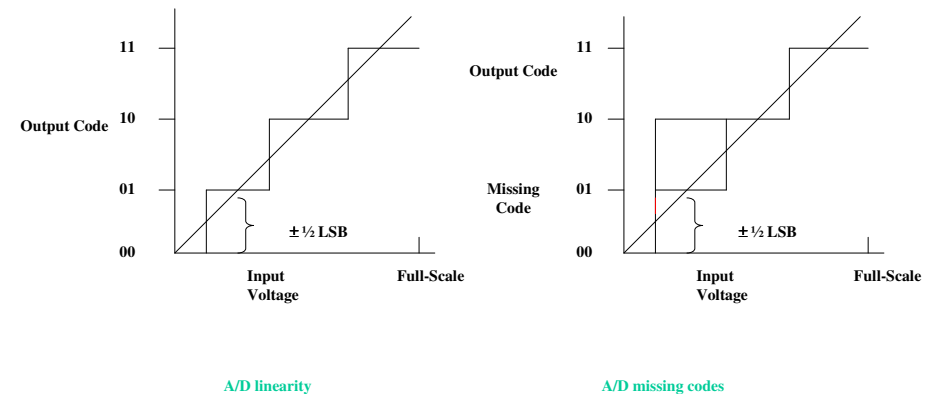
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## A/D Converter Specifications (Cont.)

- Missing codes.
  - A missing code could be caused by an internal error, especially by the D/A converter in a successive approximation converter.
- Aperture time.
  - The time that the A/D converter is “looking” at the input signal.
  - It is usually equal to the conversion time.

## A/D Converter Specifications (Cont.)



## A/D Errors

- Three sources of errors in A/D conversion:
- Noise.
  - All signals have noise.
  - Need to reduce noise or choose the converter resolution appropriately to control the peak-to-peak noise.
- Aliasing.
  - The errors due to aliasing is difficult to quantify.
  - They depend on the relative amplitude of the signals at frequencies below and above the Nyquist frequency.
  - The system design should include a low-pass filter to attenuate frequencies above the Nyquist frequency.

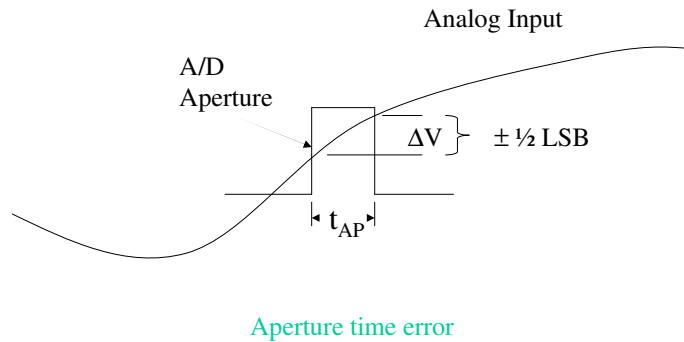
## A/D Errors (cont.)

- Aperture.
  - A significant error in a digitizing system is due to signal variation during the aperture time.
  - A good design will attempt to have the uncertainty,  $\Delta V$ , be less than one least significant bit.
  - A design equation for the aperture time,  $t_{AP}$ , in terms of the maximum signal frequency,  $f_{MAX}$ , and the number of bits in the A/D converter is

$$t_{AP} = 1 / (2 \pi f_{MAX} 2^n) \quad (4)$$

- The aperture time needed to reduce the error is surprisingly short.

## A/D Errors (Cont.)



## Reading Material

- Chapter 11: Analog Input and Output. Microcontrollers and Microcomputers by Fredrick M. Cady.
- Timers/Counters. AVR Mega64 Data Sheet.
  - PWM

## Homework

1. With the AVR lab board, connect PB7 to a LED and run the following code. What did you observe?

```
.include "m64def.inc"
.def temp=r16

ldi temp, 0b10000000
out DDRB, temp      ; Bit 7 will function as OC2.

ldi temp, 0x4A       ; the value controls the PWM duty cycle
out OCR2, temp

; Set the Timer2 to Phase Correct PWM mode.
ldi temp, (1<<WGM20)|(1<<COM21)|(1<<CS20)
out TCCR2, temp
```

## Homework

2. The A/D converter conversion time is 100  $\mu\text{s}$ . What is the maximum frequency that can be digitalized without aliasing occurring?

## Homework

3. A transducer is to be used to find the temperature over a range of  $-100$  to  $100^{\circ}\text{C}$ . We are required to read and display the temperature to a resolution of  $\pm 1^{\circ}\text{C}$ . The transducer produces a voltage from  $-5$  to  $+5$  volts over this temperature range with 5 millivolts of noise. Specify the number of bits in the A/D converter (a) based on the dynamic range and (b) based on the required resolution.