

SNS COLLEGE OF TECHNOLOGY



Coimbatore-35
An Autonomous Institution

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DEPARTMENT OF ELECTRONICS & COMMUNICATION ENGINEERING

23ECB202 – LINEAR INTEGERATED CIRCUITS

II YEAR/ III SEMESTER

UNIT 4 – ANALOG TO DIGITAL AND DIGITAL TO ANALOG CONVERTER

TOPIC 4.2 – A/D Converters & Specifications

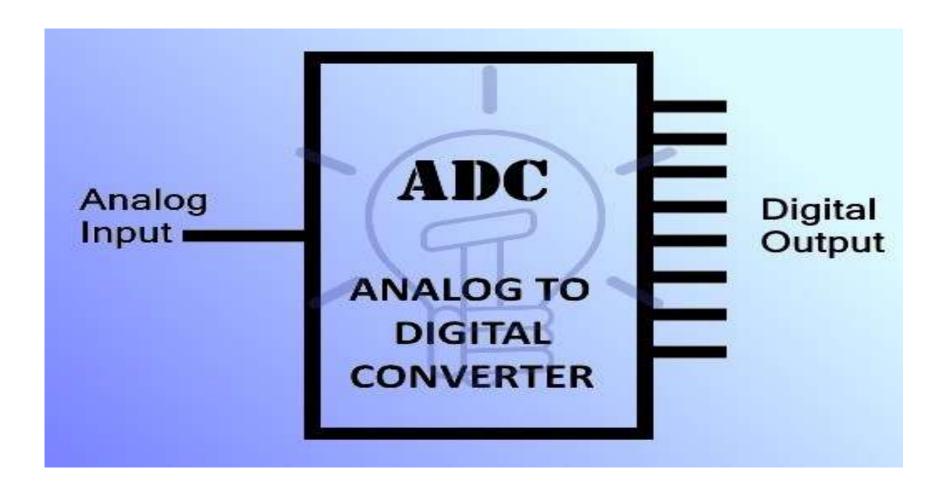




Analog to Digital Converters



- ➤ ADC stands for analog to digital converter
- > electronic device converting an analog signal into a digital signal
- The analog input signal of ADC is continuous time & continuous amplitude signal
- The output of ADC is a discrete time and discrete amplitude digital signal







Analog-to-Digital Converter



Why ADC?

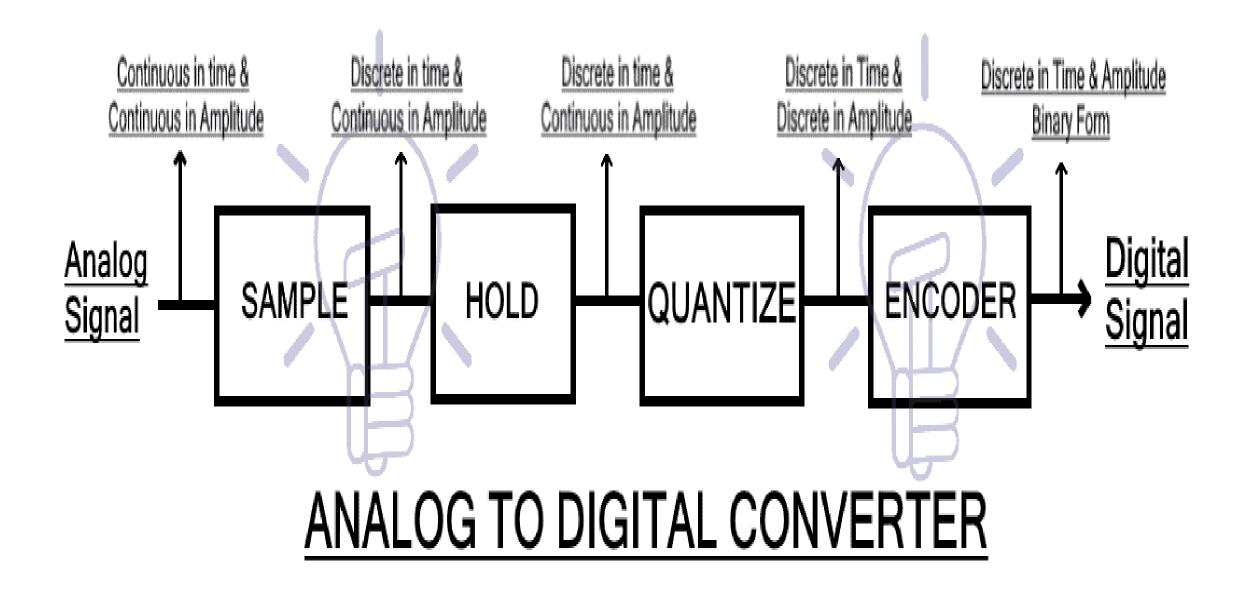
- ➤ In the real world, every real quantity such as voice, temperature, weight etc exists in the **analog state**
- it cannot be processed by any digital device such as a computer or a cell phone
- These analog quantities are converted into digital form so that a digital device can process it
- This conversion is done using **analog to digital converter**





Analog-to-Digital Converter – Block Diagram









Analog-to-Digital Converter – Block Diagram



- The analog signal is first applied to the 'sample' block where it is sampled at a specific sampling frequency
- The sample amplitude value is maintained and held in the 'hold' block
- ➤ It is an analog value
- The hold sample is quantized into discrete value by the 'quantize' block
- > Finally, the encoder converts the discrete amplitude into a binary number





Analog-to-Digital Conversion – Steps



Sample

- > sample the input analog signal at a specific time interval
- > samples are taken in continuous amplitude & possess real value but they are discrete with respect to time
- > sampling frequency plays important role in the conversion
- > maintained at a specific rate
- > sampling rate is set according to the requirement of the system.





Analog-to-Digital Conversion – Steps



Hold

- > It has no function
- > only holds the sample amplitude until the next sample is taken
- The hold value remains unchanged till the next sample

Quantize

- > This block is used for quantization
- ➤ It converts the analog or continuous amplitude into discrete amplitude
- The on hold continuous amplitude value in hold block goes through 'quantize' block & becomes discrete in amplitude
- The signal is now in digital form as it has **discrete time** & **discrete amplitude**.



Analog-to-Digital Conversion – Steps



Encoder

- The **encoder** block converts the digital signal into **binary form** i.e. into bits
- As we know that the digital devices operate on binary signals so it is necessary to convert the digital signal into the binary form using the Encoder
- This is the whole process of converting an Analog signal into digital form using an **Analog to Digital Converter**
- > This whole conversion occurs in a microsecond





Both D/A and A/D converters are available with wide range of specifications specified by manufacturer like

- Resolution
- Linearity
- > Accuracy
- Monotonicity
- Settling time
- > Stability





Resolution

- > smallest change in voltage which may be produced at the output (or input) of the converter
- Eg. an 8-bit D/A converter has 28 -1=255 equal intervals
- ➤ Hence the smallest change in output voltage is (1/255) of the full scale output range
- > Resolution can also be defined electrically, and expressed in volts
- The change in voltage required to guarantee a change in the output code level is called the least significant bit (LSB) voltage
- \triangleright The resolution Q of the ADC is equal to the LSB voltage
- The voltage resolution of an ADC is equal to its overall voltage measurement range divided by the number of intervals

$$R=rac{E_{ ext{FSR}}}{2^{M}-1},$$

Where

M is the ADC's resolution in bits

 $E_{\rm FSR}$ is the full-scale voltage range (also called 'span')





 $E_{\rm FSR}$ is given by

$$E_{\rm FSR} = V_{\rm RefHi} - V_{\rm RefLow}$$

where

 $V_{\rm RefHi}$ and $V_{\rm RefLow}$ are the upper and lower extremes, respectively, of the voltages that can be coded.

Normally, the number of voltage intervals is given by

$$N = 2^M - 1$$
,

Where

M is the ADC's resolution in bits (one voltage interval is assigned in between two consecutive code levels)





The following table 1 shows the resolution for 6 to 16 bit DACs

S.No.	Bits	Intervals	LSB size (% of full-scale)	LSB size (For a 10 V full-scale)
1.	6	63	1.588	158.8 mV
1	1			
3.	10	1023	0.0978	9.78 mV
4.	12	4095	0.0244	2.44 mV
5.	14	16383	0.0061	0.61 mV
6.	16	65535	0.0015	0.15 mV





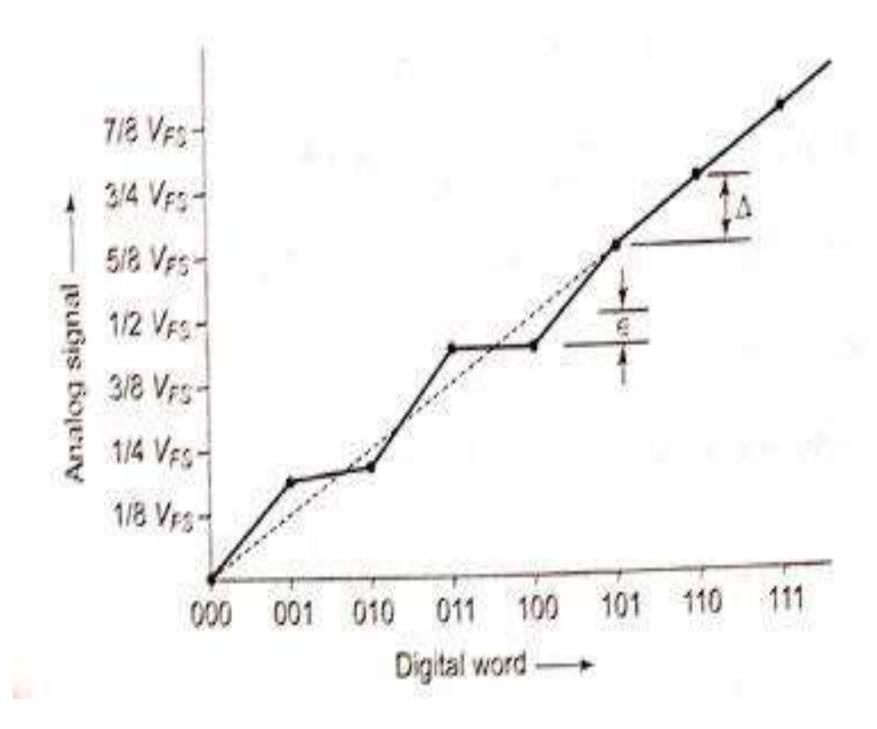
Linearity

- The linearity of an A/D or D/A converter is an important measure of its accuracy
- In an ideal DAC, equal increment in the digital input should produce equal increment in the analog output and the transfer curve should be linear
- In an actual DAC, output voltages do not fall on a straight line because of gain and offset errors as shown by the solid line curve
- The static performance of a DAC is determined by fitting a straight line through the measured output points





Linearity error of a 3-bit D/A converter







- The linearity error measures the deviation of the actual output from the fitted line and is given by ε/Δ
- The error is usually expressed as a fraction of LSB increment or percentage of full scale voltage
- \triangleright A good converter exhibits a linearity error of less than $\pm (1/2)$ LSB

Accuracy

- Absolute accuracy is the maximum deviation between the actual converter output and the ideal converter output
- ➤ Relative accuracy is the maximum deviation after gain and offset errors have been removed





Monotonicity

- A monotonic DAC is the one whose analog output increases for an increase in digital input
- A monotonic characteristics is essential in control applications, otherwise oscillations can result
- > If a DAC has to be monotonic, the error should be less than $\pm (1/2)$ LSB at each output level

Settling time

- Settling time represents the time it takes for the output to settle within a specified band $\pm(1/2)$ LSB of its final value, after the change in digital input
- ➤ It should be as small as possible
- > Settling time ranges from 100 ns to 10 μs depending on word length and type of circuit used.





Stability

- The performance of converter changes with temperature, age and power supply variations
- So all the relevant parameters such as offset, gain, linearity error and monotonicity must be specified over the full temperature and power supply ranges





THANK YOU