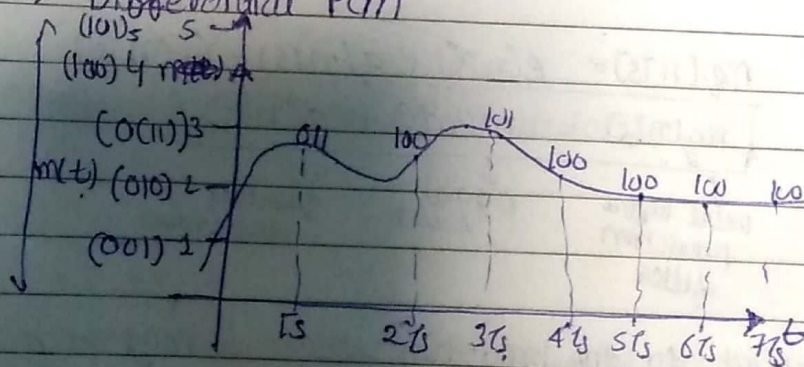


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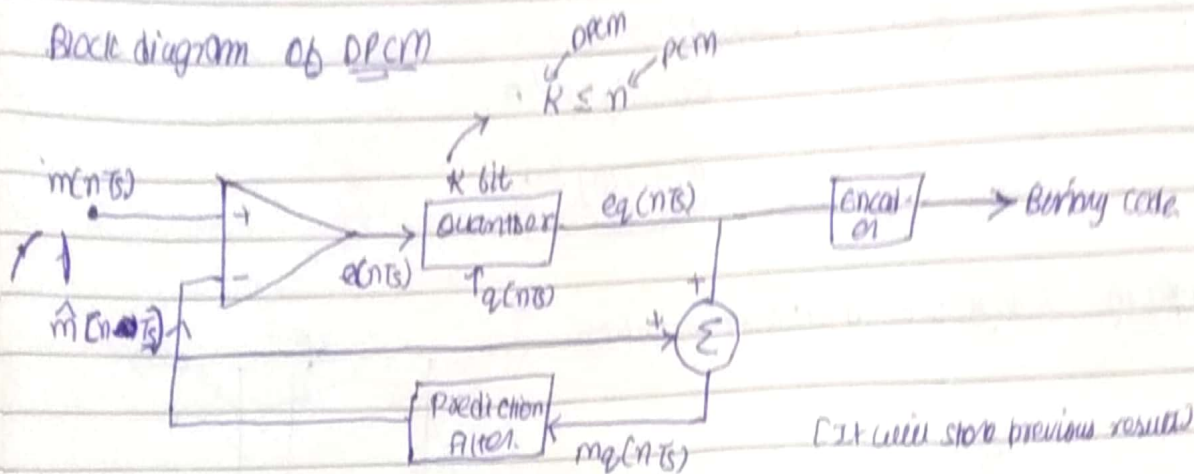
DPCM → Differential PCM



A code is assigned to each sample

→ From the figure, that at sampling instance $4T_s$ to $5T_s$, $6T_s$, $7T_s$ the same code is transmitted which is 100 so PCM does not utilise given bandwidth efficiently. Hence here we need to have a concept of differential PCM which first check the amount of information variation and then assign a ^{code} of ~~code~~ variable length.

Block diagram of DPCM



$$\rightarrow e(nTs) = m(nTs) - \hat{m}(nTs) \quad (1)$$

$$e_q(nTs) = e(nTs) + q(nTs) \quad (2)$$

$$m_q(nTs) = e_q(nTs) + \hat{m}(nTs) \quad (3)$$

$$m_q(nTs) = e(nTs) + q(nTs) + \hat{m}(nTs)$$

$$\boxed{m_q(nTs) = m(nTs) + q(nTs)}$$

\downarrow value before prediction filter
 \downarrow original value
 \downarrow quantized value (noise)

from (1) $e(nTs) + \hat{m}(nTs) = m(nTs)$

\rightarrow The input to the prediction filter is very much dependent on samples value and quantisation error. The prediction filter works in such a manner so as to minimize the error.

Prediction filter: It is having a delicate process and its memory. To maintain the data base for all sample.

The no. of bits assigned to a signal depends on the difference between $m(nTs)$ and $\hat{m}(nTs)$. More the difference, more bits will be assigned. Hence prediction filter tries to predict the next incoming sampled value.



$m(nT_s)$ and $\hat{m}(nT_s)$ have same nature

Prediction filter have the memory and processor.

→ Prediction filter store all the past values in its database and on this basis this filter tries to find out the next sampled value of signal.

→ If the past result is indicating a +ve error then $\hat{m}(nT_s)$ is given a small +ve increment in such a manner so that error $e(nT_s)$ is minimized.

→ It is having a cord system in which no of bits are less than that of PCM. $\boxed{R < n}$ $\xrightarrow{\text{PCM}}$ DPCM

★ Signal to Noise Ratio for Δ modulation ★

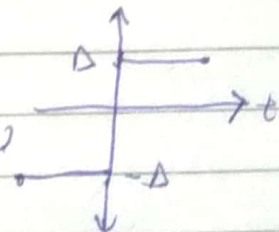
Delta modulation:

1) It is the type of DPCM

2) Here 1 bit is transmitted for each sample and sampling is done at a faster rate.

3) It is also having one bit quantiser.

4) It is also called as chord limiter. [value is predefined]



$e(nT_s) \rightarrow +ve$ (non zero bit +ve)

Δ

then assigned a value Δ

$\rightarrow 1$

Δ is a (step size)

(non zero bit -ve)

then assigned a value $-\Delta$

$\rightarrow 0$

$\rightarrow 1$ bit

$\Delta = 0$ [no transfer of data]

\downarrow negligible



one bit quantizer \rightarrow hard limiter

low pass filter / integration / delay filter

