Thansformation of analog filters into equivalent digital filters:

Digitization of analog filter mos

- 1. Approximation of derivatives method
- 2. Impulse Invariant method
- 3. Bilineal Transformation method
- 4. matched 2-transform method.

For the transformation (or) convertion to be effective, it should possen the following desirable properties:

- 1. The jr axix (ie imaginary axis) in the S-plane Should map into the unit circle in the Z-plane. Thus, there will be a direct helationship between the two frequency variables in the two domain.
- 2. The left-half of the s-plane should map interior of the unit-circle in the z-plane. Thus, a stable analog filter will be converted to a stable digital filter.

 8-plane z-plane

on jur axis on the unit circle

LH of jraxis inside unit circle

RH of jraxis outside unit circle

Impulse Invariant Transformation:

In this method, IIR digital filter is designed Such that the unit-impulse response Lan is the sampled version of the impulse response of analog filter. The main idea behind this technique is to preserve the frequency response characteristics of analog filter.

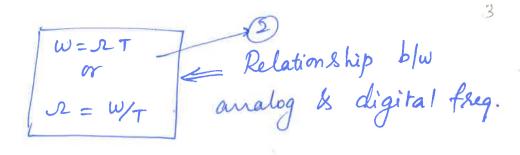
WKt
$$H_{a(s)} = \int_{-\infty}^{\infty} f_{a(t)} \cdot e^{-st} dt$$

$$H(z) = \int_{n=0}^{\infty} f_{a(t)} \cdot e^{-st} dt$$

leelationship b/w & & Z-transform is, both & & Z are complex quantities.

where $S = \sigma + j \Sigma$, $Z = M e^{j w}$ (polar form) $\sigma \to S$ seal part $S \to S$ radius of the circle in $S \to S$ phase angle $S \to S$ imaginary part. $S \to S$ phase angle

Sub $18 \times 10^{\circ}$, $e^{j\omega} = e^{(0+jx)T}$ $1 = e^{(0)T}$ $1 = e^{(0)T}$ $1 = e^{(0)T}$

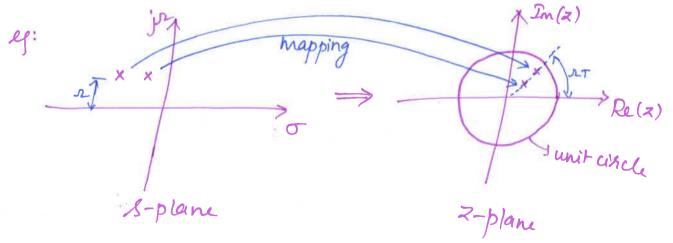


 $Z = \mathcal{L} e^{j\omega} = 121 LZ$ $|Z| = \mathcal{L} = e^{j\omega}$ $|Z| = \mathcal{L} = e^{j\omega}$ $|Z| = \mathcal{L} = e^{j\omega}$ Relationship blue analog and digital filter poles. $|Z| = \mathcal{L} = \mathcal{L} = \mathcal{L}$ and digital filter poles.

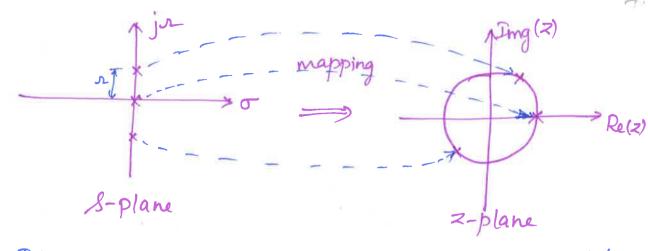
From the above equation 3, the following observations can be made:

1. If $\sigma < o$ (4. σ is negative), then analog pole's' lies on Left-half of Jz axis of S-plane.

In this case, |z| < 1, hence the corresponding digital pole's' will at lie inside the unit-circle in 2-plane.



2) If T=0 (ie real part is zero), then analog pole 's' lies on imaginary axis of s-plane. In this case |z|=1, hence the corresponding digital pole z' will lie on the unit-circle in z-plane.



3. If \$\sigma>0, (\hat{u} \sigma possitive), then analog pole is lines on Right half of jr axis of -8 plane. In this case |2| >1, hence the corresponding digital pole will lie outside the unitciscle in 2-plane

Given
Halk) = LT {-halt)}

-> h(n)= ha(t)/ t=nT T=1sec, H(2)= 7-7 {-h(n)}

Let halt) -> impulse lesponse of analog filter

Hals) -> Transfer function " "

h(n) -> Impulse lesponse of digital filter

H(z) -> Transfer function " "

$$H_{a}(s) = \underbrace{\frac{C_{k}}{s-p_{k}}} = \underbrace{\frac{C_{1}}{s-p_{1}}} + \underbrace{\frac{C_{2}}{s-p_{2}}} + \dots \xrightarrow{F_{2}} \underbrace{F_{2}}$$

$$ha(t) = \underbrace{\sum_{k=1}^{C_{k}} c_{k}^{P_{k}t}}_{K=1}, \quad t \ge 0$$

$$= c_{1}e^{P_{1}t} u(t) + c_{2}e^{P_{2}t} u(t) + \dots$$

$$= c_{n}e^{P_{n}t} u(t) + c_{n}e^{P_{n}t} u(t) + \dots$$

$$h(n) = h_a(t)/t = nT = h_a(nT)$$

$$fina(nT) = \sum_{k=1}^{N} c_k e_{k}^{p_k nT} \qquad 6a$$

$$= c_1 e_{ua(nT)}^{p_1 nT} + c_2 e_{ua(nT)}^{p_2 nT} + c_3 e_{ua(nT)}^{p_3 nT} + c_4 e_{ua(nT)}^{p_3 nT} + c_5 e_{ua(nT)}^{p_3 nT} + c_6 e_{ua(nT)}^{p$$

Take z-transform of egn 6.

$$H(2) = \frac{N}{k=1} \frac{C_{k}}{1-e^{P_{k}T}-1} \sum_{n=0}^{\infty} a^{n} = \frac{1}{1-a}, |a| < 1$$

(Or)

27 of egn 6b

$$H(z) = \frac{C_1}{1 - e^{P_1 T_2 - 1}} + \frac{C_2}{1 - e^{P_2 T_2 - 1}} + \cdots$$

hence If
$$Ha(s) = \frac{S}{S-P_k} \frac{C_k}{S-P_k}$$
 then $H(z) = \frac{S}{k=1} \frac{C_k}{1-e^{P_kT}z^{-1}}$

For high sampling sate (small T), digital filter gain is high. Therefore, instead of @ , we can me

$$H(z) = \sum_{k=1}^{N} \frac{T \cdot C_k}{1 - e^{P_k T} - 1}$$

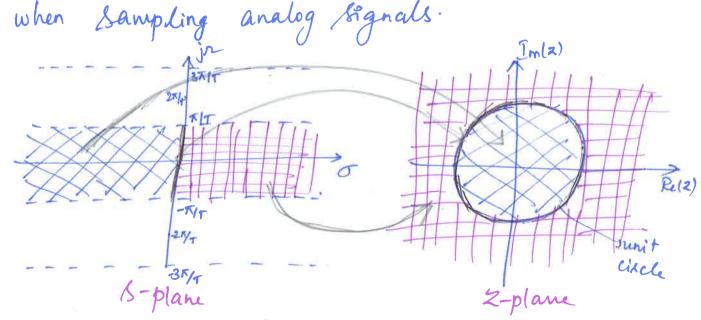
drawback of Impulse Invariant method:

If analog poles are having same Ireal pasts and imaginary parts that and differ by some integer multiple of $\frac{2\pi}{T}$, then

they are mapped to a single p digital pole in 2-plane. for eg: let us consider two poles SI= O#+ j2 $S_2 = \sigma + j\left(x + \frac{2\pi}{T}\right)$ $Z_1 = e^{S_1 T} = e^{(\sigma + j x)T} = e^{\sigma T} = e^{\sigma T}$

 $Z_2 = e^{S_2 T} = \int_{0}^{T} (x + 2\pi) T$ = et ejet [ejet = 1 122=21 =Z

this impulse invariant technique results in many-to-one-mapping. This is due to aliasing



Foron the above figure, we can say that, the Strip of width 27/7 in the S-plane for

8.

Values of s in the sange $-\frac{\pi}{T} \leq 2 \leq \pi/T$ is mapped into the entire 2-plane. Similarly, the strip of whoth $2\pi/T$ in the s-plane for values of s in the range $\pi/T \leq 2 \leq \frac{3\pi}{T}$ is also mapped into entire-2-plane. Likewise, the strip of whoth $\frac{2\pi}{T}$ in the strip are for values of s in the trange $-\frac{3\pi}{T} \leq 2 \leq -\frac{\pi}{T}$ is also mapped into the entire 2-plane.

In general, any Strip of width $2\pi/T$ in the S-plane for values of S in the range $(2k-1)\frac{\pi}{T} \leq x \leq (2k+1)\frac{\pi}{T}$ (where kix an integer) is mapped into the entire z-plane.

Due to the presence of aliasing, the impulse invariance method is appropriate for the design of LPF & BPF only. This method is unsuccessful for implementing digital filters such as a HPF.

For analog transfer function $H(S) = \frac{2}{(S+1)(8+2)}$, determine H(z) using Impulse Invariant method for i) T = 1 sec ii) T = 0.18ec

Soln: Given: Hals) =
$$\frac{2}{(s+1)(s+2)}$$

By using Partial Fraction Expansion, method,

$$=\frac{2}{(3+1)(3+2)}\Big|_{S=-1}=\frac{2}{-1+2}=\overline{2}=\overline{A}$$

$$= (8+2) \cdot \frac{2}{(8+1)(8+2)} = \frac{2}{-2+1} = [-2=8]$$

$$|Ha|(S) = \frac{2}{S+1} + \frac{-2}{S+2} = \frac{2}{S-(-1)} + \frac{-2}{S-(-2)}$$
By impulse invariant transformation, P_1

$$P_2$$

$$H(z) = \frac{2}{1 - e^{-7}z^{-1}} + \frac{-2}{1 - e^{-27}z^{-1}}$$

$$H(z) = \frac{2}{1 - e^{-1}z^{-1}} + \frac{-2}{1 - e^{-2}z^{-1}}$$

$$H(z) = \frac{2}{1 - 0.3678 z^{-1}} + \frac{-2}{1 - 0.1353 z^{-1}}$$

$$= \frac{2(1 - 0.1353 z^{-1}) - 2(1 - 0.3678 z^{-1})}{(1 - 0.3678 z^{-1})(1 - 0.1353 z^{-1})}$$

$$H(z) = \frac{0.465 z^{-1}}{(1 - 0.3678 z^{-1})(1 - 0.1353 z^{-1})}$$
Annuer

homeworks: for T=0.1sec.

Phm2: Using impulse invariant method, determine

$$H(2)$$
 if $H(3) = \frac{1}{(S+1)} (S^2 + S + 1)$, Assume $T = 1 \text{ sec}$

Soln:

 $A(3) = \frac{A}{(S+1)} + \frac{B}{(S+0.5 + j \circ .866)} + \frac{S^2 + S + 1}{(S_{1,2} = -1 \pm \sqrt{1-4})} + \frac{C}{(S+1)(S^2 + S + 1)} + \frac{C}{(S+1)(S^2 + S + 1)} + \frac{C}{(S+1)(S+3+1)} + \frac{C}{(S+1)(S+3+1)} + \frac{C}{(S+1)(S+3+1)} + \frac{C}{(S+1)(S+3+1)} + \frac{C}{(S+1)(S+3+1)(S+3+1)} + \frac{C}{(S+1)(S+3+1)(S+3+1)} + \frac{C}{(S+1)(S+3+1)(S+3+1)(S+3+1)} + \frac{C}{(S+1)(S+3+1)(S+3+1)(S+3+1)(S+3+1)} + \frac{C}{(S+1)(S+3+1)(S+3+1)(S+3+1)(S+3+1)(S+3+1)} + \frac{C}{(S+1)(S+3+1)$

 $= 70.52 + \frac{1}{(-j1.732)(0.5-j0.866)} = \frac{1}{-j0.866-1.5}$

$$B = \frac{-1.5 + j \cdot 0.866}{-1.5 + j \cdot 0.866}$$

$$= \frac{-1.5 + j \cdot 0.866}{(-1.5)^{2} + (0.866)^{2}} = \frac{-1.5 + j \cdot 0.866}{3}$$

$$B = -0.5 + j \cdot 0.288$$

$$C = B^{*} = -0.5 - j \cdot 0.288$$

$$H(8) = \frac{1}{(s+1)} + \frac{-0.5+j0.288}{(s+0.5+j0.866)} + \frac{-0.5-j0.288}{(s+0.5-j0.866)}$$

$$P_{1} = -1, \quad P_{2} = -0.5-j0.866, \quad P_{3} = -0.5+j0.866$$

Using impulse invariant technique,

$$H(2) = \frac{1}{1 - e^{-j} 2^{-1}} + \frac{-0.5 + j0.288}{1 - e^{-0.5} - j0.866} + \frac{-0.5 - j0.288}{1 - e^{-0.5} + j0.866}$$

$$= \frac{-0.5}{e} = \frac{-j0.866}{e} = \frac{e^{-0.5} + j0.866}{e^{-0.5} + j0.866}$$

(OS(0.866)-j&in(0.866) (in Ladian mode)

$$H(z) = \frac{1}{1 - 0.368z^{-1}} + \frac{-1 + 0.66z^{-1}}{1 - 0.786z^{-1} + 0.368z^{-2}}$$
Answer

home work 28 384

2. Convert the analog filter with transfer function $Ha(S) = \frac{Sto.1}{(Sto.1)^2 + 9}$ into a digital IIR filter by means of impulse invariance method.

3.
$$Ha(s) = \frac{1}{s^2 + \sqrt{2}s + 1}$$
, $T = isc$

Bilineal Transformation method:

The bilinear transformation is a conformal mapping that transforms the jor axis into the unit-Circle in the z-plane only once, thus avoiding aliasing of frequency components.

Furthermore, all points in the LH of Splane are mapped inside the uniteixcle in z-plane and all points in the RH of S-plane are mapped into corresponding points outside the unit-circle in Z-plane.

Let us consider an analog linear filter with System function,

$$H(S) = \frac{b}{S+a} = \frac{Y(S)}{X(S)} \longrightarrow 0$$

 $S \cdot Y(S) + a \cdot Y(S) = b \cdot X(S)$ $\longrightarrow ②$ Taking Inverse Laplace transform of eqn ②,

$$\frac{dy(t)}{dt} + a.y(t) = b.x(t) \longrightarrow 3$$

If we approximate y(t) by trapezordal formula,

where y'(t) -> derivative of y(t).

The approximation of the integral in equal by the trapezoidal formula at t=nT & to=nT-T yields,

$$y(nT) = -\frac{T}{2} \left[y'(nT) + y'(nT-T) \right] + y(nT-T) \longrightarrow 5$$

from eqn (3), at t=nT

$$y'(n\tau) = -a y(n\tau) + bx(n\tau)$$

Substitule egn 6 in egn 6,

$$(a) \Rightarrow y(nT) = \frac{T}{2} \left\{ -ay(nT) + bz(nT) - ay(nT-T) + bz(nT-T) \right\}$$

+ y (nT-T) -> 7 y'(nT)

which Implies,

$$y(nT) + \frac{aT}{2}y(nT) - \left[1 - \frac{aT}{2}\right]y(nT-T) = \frac{bT}{2}\left[x(nT) + x(nT-T)\right]$$

With $y(n) = y(n\tau)$ & $x(n) = x(n\tau)$ $y(n\tau - \tau) = y(n-1)$ & $x(n\tau - \tau) = x(n-1)$

We obtain, $(1+\frac{aT}{2})y(n) - (1-\frac{aT}{2})y(n-1) = \frac{bT}{2} \left[x(n) + x(n-1)\right]_{2}$

Taking z-tlansform of egn (5),

$$\left(1+\frac{a\tau}{2}\right)\gamma(2)-\left(1-\frac{a\tau}{2}\right)z^{-1}\gamma(2)=\frac{b\tau}{2}\left(1+z^{-1}\right)\chi(2)$$

hence, the transfer function of digital

filter is,
$$H(z) = \frac{Y(z)}{X(z)} = \frac{bT_2(1+z^{-1})}{1+\frac{aT}{2}-(1-\frac{aT}{2})z^{-1}}$$

$$H(z) = \frac{bT/2(1+z^{-1})}{(1-z^{-1}) + \frac{aT}{2}(1+z^{-1})}$$

Dividing numerator & denominator of eqn (16) by $T_{2}(1+2^{-1})$, we get

$$H(a) = \frac{b}{\frac{2}{T}(\frac{1-z^{-1}}{1+z^{-1}}) + a}$$

If
$$H(S) = \frac{b}{S+a_1}$$
 then $H(z) = \frac{b}{T(\frac{1-z^{-1}}{1+z^{-1}})+a}$

S is mapped
$$\frac{2}{t}\left(\frac{1-z^{-1}}{1+z^{-1}}\right) \longrightarrow 2$$

$$S = \frac{2}{T} \begin{bmatrix} 1-z^{-1} \\ 1+z^{-1} \end{bmatrix} = \frac{2}{T} \frac{z^{4}[z-1]}{z^{4}[z+1]}$$

$$=\frac{2}{T}\begin{bmatrix}\frac{2-1}{2+1}\end{bmatrix}=\frac{2}{T}\begin{bmatrix}\frac{91e^{jw}-1}{91e^{jw}+1}\end{bmatrix}$$

$$= \frac{2}{T} \left[\frac{\text{R CoSw} - 1 + j \text{R Sinw}}{\text{R CoSw} + 1 + j \text{R Sinw}} \right]$$

$$= 2 \left[\frac{2}{(28w-1)+j? \sin \omega} \right] \left[\frac{(2\cos w+1)-j? \sin w}{(2\cos w+1)-j? \sin w} \right]$$

$$= \frac{2}{T} \left[\frac{x^2 \cos^2 w - 1 + x^2 \sin^2 w + j 2x \sin w}{(x \cos w + 1)^2 + x^2 \sin^2 w} \right]$$

$$= 9^2\cos^2\omega + 1 + 2\pi\cos\omega$$

$$S = \frac{2}{T} \left[\frac{9^2 - 1}{1 + 9^2 + 2900w} \right] + j + \frac{2}{T} \left[\frac{27 \sin w}{1 + 9^2 + 2900w} \right] - \sqrt{9}$$

Compare egn (8) & (3b) & (4)

$$\sigma = \frac{2}{T} \begin{bmatrix} 9^2 - 1 \\ 1 + 9^2 + 29 \cos \omega \end{bmatrix}, \quad \boxed{15}$$

$$\Omega = \frac{2}{T} \left[\frac{291 \, \text{Sinw}}{1 + 8^2 + 29 \cos \omega} \right] \rightarrow 16$$

From the Iquations (5) & (6), we can say that if 9<1, then 0<0, where 9=121 (radius of the circle in 2-plane)

ie LH of jn axis] mapped intide the unit-circle of S-plane poles Jinto & in the z-plane

if 9>1, & then 0>0,

ie. RH of Je axis Jare mapped southide the unit circle of s-plane poles of in the z-plane.

If 9=1, ther =0 and

$$\Lambda = \frac{2}{T} \frac{2 \sin \omega}{2 + 2 \cos \omega} = \frac{2}{T} \frac{\sin \omega}{1 + \cos \omega}$$

$$SL = \frac{2}{T} \tan \left(\frac{\omega}{2}\right)$$
(0r)

$$w = 2 \tan^{-1} \frac{xx}{2}$$

gives the relationship between analog and and digital filter frequencies.