# DIGITAL PROCESSING FOR SPEECH

11 7.7 W(Z) = Z brz-1-Zab2-k

a) For w/z) be suitable for this application, w/z) should be stable. All the poles must be invide the unit citle. It should be of the form of Low pass filter with frequency on frequency domain at w=0.

1- Zap2-k

 $X_{n}(e^{j\omega}) = \overline{Z}_{n}(m) tol(n-m), e^{-j\omega m}$ which can be written as

= x[m] ejwn & w[n]

Now  $x_n(e^{j\omega}) = z(n)e^{-j\omega}n_{\omega}\omega(n)$ 

 $\chi_{n}(e^{i\omega}) = \chi(z) e^{-i\omega n} \frac{\omega(z)}{\sum_{j=0}^{N} \int_{z_{j}} z_{j}}$   $= \chi(z) e^{-i\omega n} \frac{\sum_{j=0}^{N} \int_{z_{j}} z_{j}}{\sum_{j=0}^{N} \int_{z_{j}} z_{j}}$ 

$$\begin{array}{l} \chi_{n}(e^{j\omega}) \left(1 - \frac{Np}{2} a_{0} z^{-k}\right) = \chi(z) e^{-j\omega n} \frac{Np}{2} b_{n} z^{-n} \\ \chi_{n}(z) \left(1 - \frac{Np}{2} a_{0} z^{-k}\right) = \chi(z) e^{-j\omega n} \frac{Np}{2} \sum_{r=0}^{Np} \sum_{$$

101 <1 for ordalitity and a should be real.  $\omega(e^{\pm j\omega}) =$ ١- ٥ ((١٥) - ن ٥ ١٥ (١٥) 1-a cos(w) + a; sin (w) Lets take minutuale equared frequency response-[v(eiw) 2 (1-aus(w)) + azsuz (w). 1-2a cas(w) + a2 cay2(w) 1 - 2a cos(w) + a2 we define the cutoff prequency of system for which; | w ( eiwc) | 2 = 1 | w(ei 0) | 2 Frequency resolutions is 100Hz on samplingrate of 10kHz. Since cut of frequency is half of 100Hz = [100 = 50Hz 27 Je/ = 27 50 = T/100 =  $\left|\omega\left(e^{j\omega c}\right)\right|^2 =$ 1 (n(e;0))2 = 1 1-102-20x60) ne prom [n (62 nc) 13 = [ 1 n (620) ] 5

 $= \frac{1}{17a^2 - 2a\cos(7/100)} = \frac{1}{2(1+a^2 - 2a)}$ 2+202 - 40 = 1+02 - 20 con( 1/100) 1 + a2 - 4a + 2a cos/1/100) = 0 1+2 - 2a (2- cos (1/00)) =0 1-12 - 2a (2-0.9995) =0 14 a2 - 2a (1.00049) =0 1+a2-2.00098 a=0 a= 1.031798 and a=0.96918 sirce 19161 me house a= 0.96918 2) x,(e sw) = 2 x(m) w(n-m) evmu mcn) = vb, ntv) a) X 56 (e' w) barreally will use x(m) with m less than or equal to 49 else there will

7,00 (eJw) will us x(m) with m & 100

elses there will be aleasing

be always

$$|x| \times (e^{j\omega}) = \sum_{n=-\infty}^{\infty} \omega(n-n) \times (n) e^{-j\omega n}$$

$$|x| = x (n) = x (e^{j\omega})$$

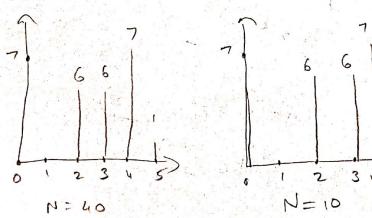
$$|y(n)| = x (n) = x (n)$$

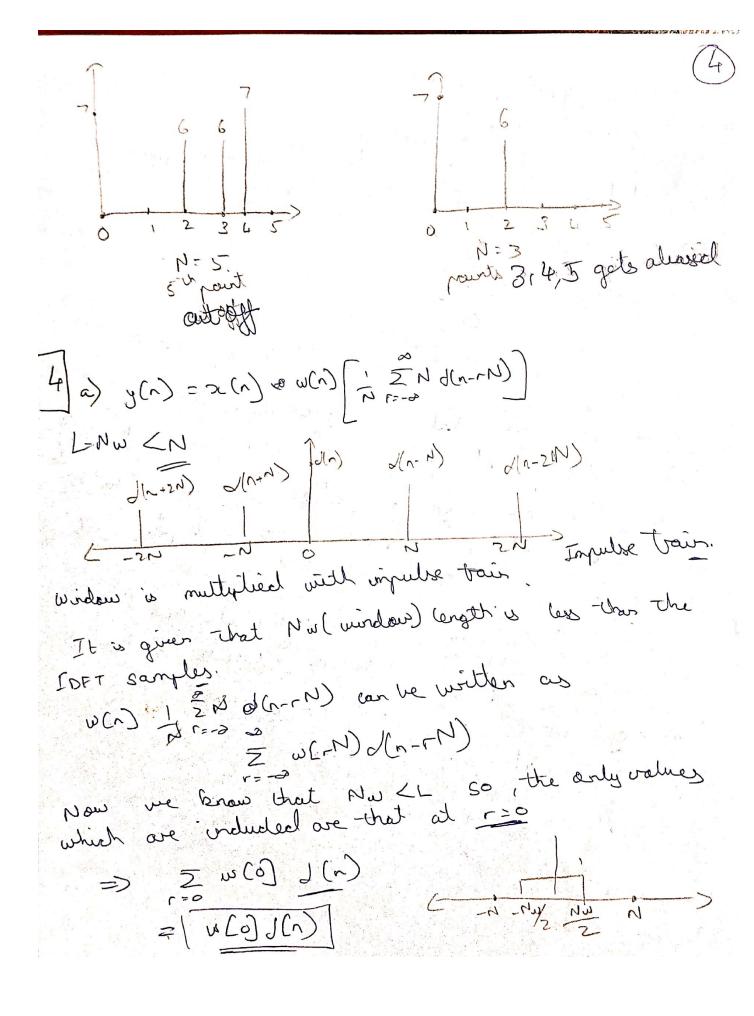
$$|y(n)| = x (n) = x (n)$$

$$|y(n)| = x (n) = x (n) = x (n)$$

$$|y(n)| = x (n) = x (n) = x (n)$$

$$|y(n)| = x (n) = x (n) = x (n)$$





Now,  $y(n) = x(n) = w(n) | \sum N d(n-n)$  y(n) = x(n) = w(n) | d(n)This is I only at one position convolution changes to multiplication. y(n) = x(n) = w(n) y(n) = x(n) = w(n) y(n) = x(n) = y(n) y(n) = x(n) = y(n)

H) Now we know that w(n)=0 at n= in, ± 2nd...

This implies that w(n) is non zero only at w(o)

=> Any window size no matter of what length
will have only one non - zero sample that'y

at n=0, ie, w(o)

=> Since only on term occurs at n=0, so

any window length will not course any
any window length will not course any
alwaying and we can construct the

refert signal back again.

(1) b(0) u(n) [= 0(0) d(n) - 0(0) d(n) Taking DTFT on both sides LHS = Z w(n) (= (n-rN) e-jw) Σω(n) e-jwn = ω(ω) given = w (w) = d (n-rN) we can write  $\frac{2}{5}d(x-rN)$  as  $\frac{1}{5}e^{\pm i\frac{2\pi kn}{N}}$ = \(\int \omega N-1 22 W (N) e-3 (W 42 T/2) n 1 N=7 Z W (N) e-3 (W 42 T/2) n N  $w \in \text{Enou} \quad \stackrel{\text{de}}{\underset{n=-\infty}{\text{de}}} \quad \omega(n) \in \text{de} \quad \omega(w)$  $= \sqrt{\frac{2}{2}}\omega(n)e^{-\sqrt{2\pi}k} - 2\pi k$ => LHS=1Z W(w-2 1k) Nb=0 RHS = \frac{1}{2} \omega (0) \d(n) \equiv \frac{1}{2} \omega^2 we know Z d(n)e jun RHS = w (0).1 = w (0)

Now LHS = R+15 N-1 N-1 N b=0 (w-27b) = W(o) for any w 5) a) w"(r) = w(+R-2) downsampled version of we know  $\hat{\omega}$  (e) =  $\frac{1}{R}$   $\frac{1}$ This is vaircally sampling around the nit circle in frequency domain.

if we keep w = 0  $= \sum_{k=0}^{\infty} w(e^{j2\pi 7/k})$ There is a coul of rossible time domain aliasing, which can happen this needs to he avoided. From the textbook by Robiner. time alianing occurs when window length is way greater than sampliffer length This is the coes where obiaising occursion time domain, where contains, samples are lost. So, me know to avoid this undow tength must be length samples => Nw or L2N/ => [R < L < N] from text book

RSL shalled be the case for undereampling the first time book as it a nothing but frequency domain sampling. WC = 27 - The condition, if we consider the case of down sampling. =) For (w) = zrr we know that From the constraint conditions => 2 Tr 7, 27 L REL

$$V_{r}(k) = 4r (e^{j\omega k}) = x_{r}R (e^{j2\pi ky/N})$$

$$V_{r}(m) = \chi(m) \quad \text{with } W(rR-m)$$

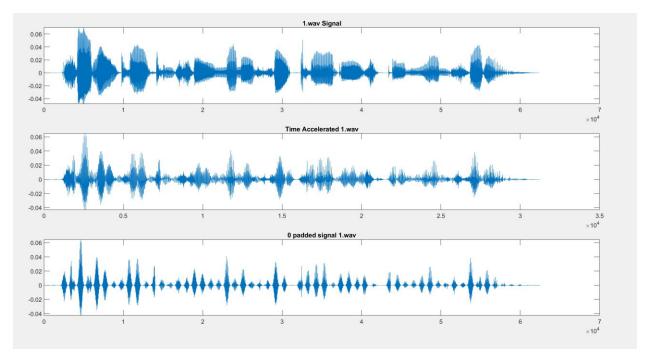
$$V_{r}(m) = \chi(m) \quad \text{with } W(rR-m)$$

$$V_{r}(m) = \chi(n) \quad \text{with } W(rR-n)$$

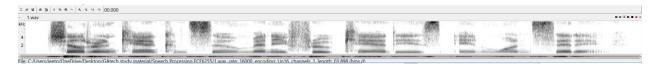
$$V_{r}(m) = \chi(n) \quad \text{with } W(rR-$$

5) I have used a 1024 pt FFT and (7) Jundow length of 1624. On downsampling and reforming idst, we can hear a lot gapes in time due to which we hear staggered sound: Sound is heard and is understandable as we remove only alternate samples and most of the date still remains. Now, in case of interpotation outh zero padding, we add a conschol outh zero padding, we add a conschol outland the high frequency and which means components are replaced by a which means putch is being was orbit intially at 27 the prequency was orbit intially at 27 we downsampled it to Tr. So, frequency we downsampled it to Tr. So, frequency we have hopens and lots of low suppression hoppens and lots of low Jeguney components come up. The voice file, is heard that we can still undertand but the voice is robolie sound, we didn't remove all the date but only halfed the date; So the time indesc is relatively short (ode and outputs are stored and uploaded with submission.

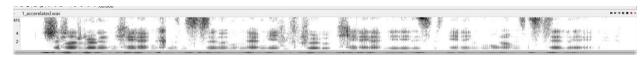
#### 1.wav



## Original signal



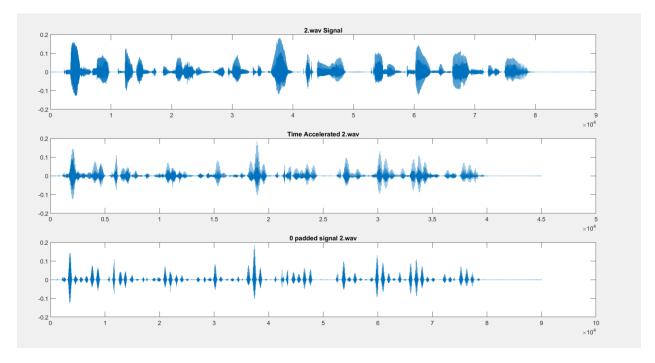
#### Time accelerated signal



# Time interpolated signal



#### 2.wav



# Original signal



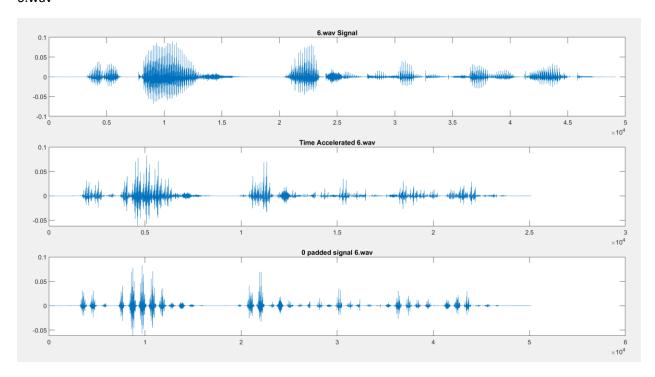
#### Time accelerated signal



#### Time interpolated signal



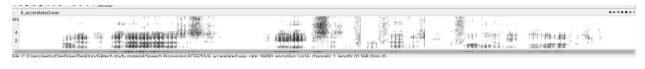
#### 6.wav



## Original signal



### Time accelerated signal



#### Time interpolated signal

