## SAMPLING THEOREM

## CODE:

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% sampling theorem verification
%right sampling fs=2fm(fs is sampling frequency and fm is max frequency)
%over sampling fs>4fm
%under sampling fs<2fm aliasing
clc;
t=0:0.01:0.2;
f1=input('enter frequency 1: ');
f2=input('enter frequency 2: ');
xa=cos(2*pi*f1*t) + cos(2*pi*f2*t);%continuous signal
fm = max(f1,f2);
% right sampling
fs=2*fm;
%t=nT i.e continous to dectetrrtr time
ts=1/fs;%sampling interval
n = 0:1:0.2*fs;% discrete scale
xd = cos(2*pi*f1*(n*ts)) + cos(2*pi*f2*(n*ts));% discrete time signal
%xa is converted to xd by taking samples at particular intervals of time
%thus give us dt signal
subplot(3,1,1)
plot(t,xa);
xlabel('t in seconds');ylabel('x(t)');title('analog signal');hold on;
stem(n*ts,xd);%to put points on the ct signal
hold off;
subplot(3,1,2)
stem(n,xd);
xlabel('n in seconds');ylabel('d(t)');title('analog signal of dt');
xr=xd*sinc((t-(n'*ts))/ts);
subplot(3,1,3)
plot(t,xr);
xlabel('t in seconds');ylabel('xr(t)');title('right sampling');
%interpolation to get back the xa by multiplying the sampled signal xd by sync function
%thus verification can happen if xa = xr i.e right sampling has taken place
```

