

a7_Q4_highlowencoding

May 3, 2021

1 Linear Systems and Noise Assignment 7 - Signals in noise

The following should translate fairly smoothly to Python, Matlab, etc. else, question |> me!

2 Random bitstream: high-low encoding

```
[18]: using FFTW, Plots, LaTeXStrings, Plots.Measures
      gr(grid=false, legend=false, size=(650, 300));
```

```
[3]: MHz = 1e6 # time unit for bits
      T = 1/MHz # length of one bit in seconds.
      Nt = 2^12 # fine grid for sampling
      N = 32    # number of t grid points in one bit
      dt = T/N  # time grid point spacing
      t = (0:(Nt-1))*dt # total sampling time
      Ns = 1000 # number of realizations of the bitstream to average
```

```
[3]: 1000
```

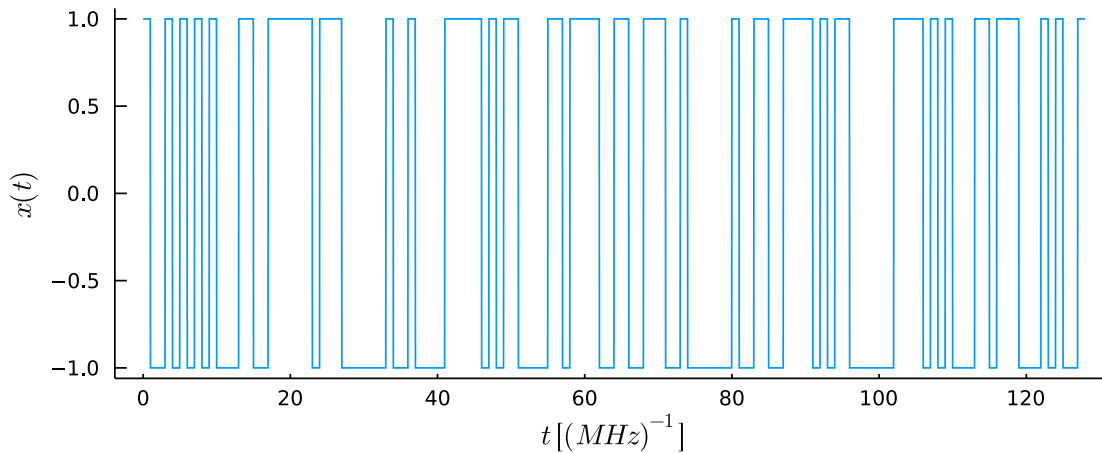
2.1 Plot one realization

Make a single random bitstream over $[0, N_t dt]$, broken into N_t/N bit intervals:

```
[4]: x = zeros(t) # create empty vector to hold each realization
      Nbits = Nt÷N # integer division
      for k in 0:Nbits-1
          bit = N*k+1:N*k+N # bit location in x(t)
          x[bit] = rand() > 0.5 ? ones(N) : -ones(N) #flip a coin (? : is shorthand
          ↪ for if else)
      end
      Phi = abs2.(dt*fft(x))/(Nt*dt); # PSD for one realization.
      # Two dt's here: first because we are approximating integral using DFT.
      # Second to normalize correctly for average power
```

```
[20]: plot(t*MHz, x); xlabel!(L"t \; [ (MHz)^{-1} ] ", bottom_margin=1cm); ylabel!(L"x(t)")
```

```
[20]:
```



Average over N_s realizations of the bitstream:

```
[21]: x = zero.(t) # zero vector to hold each realization
      Phi = zero.(t) # zero vector to accumulate average power-spectral density
      for i in 1:Ns
          # make a random bitstream over [0,Nt*dt], broken into Nt/N bit intervals
          Nbits = Nt÷N
          for k in 0:Nbits-1
              bit = N*k+1:N*k+N # bit location in x(t)
              x[bit] = rand() > 0.5 ? ones(N) : -ones(N) #flip a coin
          end
          Phi .+= abs2.(dt*fft(x))/(Nt*dt); # Accumulate PSD for Ns realizations
      end
      Phi = fftshift(Phi/Ns); # normalize and shift
```

```
[22]: nu = (-Nt÷2:Nt÷2-1)/Nt/dt;
```

```
[23]: Phi_a(nu)=T*(sinc(nu*T))^2 # Analytical PSD of bitstream, after averaging over
      ↪ random bits
```

```
[23]: Phi_a (generic function with 1 method)
```

Let's work this out. From the hint in the assignment, if we consider a finite time interval (since the signal only has finite average power, and not finite energy)

```
[24]: Phith = Phi_a.(nu);
```

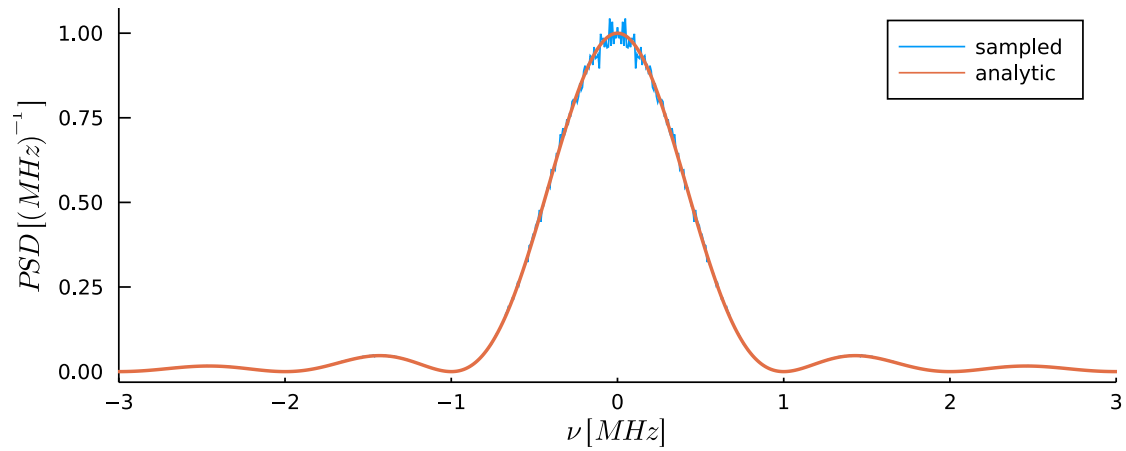
```
[25]: plot(nu/MHz,MHz*Phi,label="sampled",legend=:topright)
      plot!(nu/MHz,MHz*Phith,label="analytic",lw=2)
```

```

xlims!(-3,3);ylabel!(L"PSD\; [(MHz)^{-1}]");xlabel!(L"\nu\; [MHz] ",bottom_margin=1cm)

```

[25] :



Can be made smoother with more samples, but the essential point is clear: the PSD has its peak power at zero frequency.