ECE537: Lab 4 Report

In the first part of the lab, we will be simulating low pass Guassian, and then we will use the spectrual analysis to have a better understanding of it.

Throughout this lab, the <u>Distributions.jl</u> package in Julia has been utilized to be able to use the probability constructs in code.

using Distributions, StatsBase, StatsPlots, LinearAlgebra, LaTeXStrings, PlutoUI, DSP
, Plots

PlotlyBackend()

1. Generating Low-Pass Random Processes

Given an infinite sequence of i.i.d. Gaussian random variables, $\{X_k\}$, we can construct a bandlimited White Gaussian Noise (WGN) process by the Shannon-Nyquist sampling theorem.

$$X(t) = \sum_{-\infty}^{\infty} X_k \mathrm{sinc}\left(rac{t-kT}{T}
ight),$$

where $X_k = X(kT)$. and for this lab, we choose $X_k \sim N(0,1)$ and the summation above matches the convolution $\sum_k X(kT)\delta(t-kT)*2Bsinc(2Bt)$ for B=1/2T, implying that X(t) is a bandlimited (low-pass filtered) white noise signal with bandwidth |f| < B.

To numerically approximate,

$$X(t)pprox \sum_{k_t-m}^{k_t+m} X_k ext{sinc}\left(rac{t-kT}{T}
ight),$$

where $k_t = \lfloor t/T \rfloor$. For the purpose of this lab, we choose the approximation limit m=5 and sampling time T=1 . and we assume that X(t)=0 for t<0

X_t (generic function with 1 method)

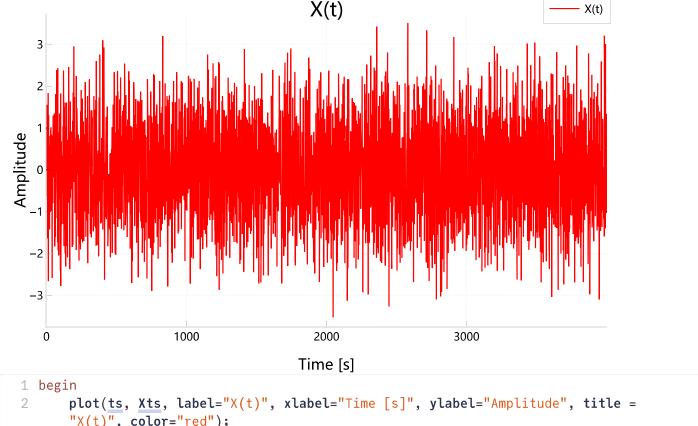
```
1 begin
 2
       # basic parameters
 3
       mu = 0;
 4
       sigma = 1;
 5
       m = 5;
 6
       T = 1;
 7
       deltaT = T / m;
 8
       # some useful functions to calculate the X(t)
9
       # calculate the X_k
10
11
       X_k(n) = [Normal(mu, sigma) for k = 1:1:n];
12
       # calculate the k_t - m to k_t + m
13
14
       k_t(t) = floor(Int, t / T);
15
       k_t(t, m) = \max(k_t(t) + m, 1);
16
17
       # calculate the sum of target values
       X_{t}(X_{k}, t, T, m) = sum([X_{k}] * sinc(t / T - k) for k = k_t(t, -m):k_t(t, m)]);
18
19 end
```

[0.0, 0.188287, 0.447049, 0.746496, 1.04299, 1.28742, 1.49414, 1.55781, 1.4545, 1.19716, 0

```
begin
    # generate the X_k array and define the true function
    Xk = rand.(X_k(20_000));
    Xt(t) = X_t(Xk, t, T, m);

# generate the t series and calculate the corresponding function value
    ts = 0:deltaT:(4000 * T);
    Xts = Xt.(ts)

end
```



X(t)

```
"X(t)", color="red");
    plot!(; xlims=(ts[1],ts[end]));
end
```

2. LTI Systems and Random Processes

Now, we consider a a linear time-invariant (LTI) filter with impulse response $h(t) = e^{-at}(u(t) - u(t-20))$, i.e., a truncated RC low-pass filter with parameter a that is cut to zero for t>20 seconds. The input to the system will be the bandlimited WGN process, X(t), produced above and the output random process, Y(t), is a modified or an approximate Ornstein-Uhlenbeck process due to the time-windowed h(t).

From analysis of deterministic signals and systems, we know that the output random process Y(t) = X(t) * h(t) which is the continuous-time convolution of the input random process and the impulse response. For computational purposes, we can approximate it as a Riemann integral,

$$Y(t)pprox \sum_{k=-\infty}^{\infty}X(t- au)e^{-a au}(u(au)-u(au-20))d au,$$

where for the purpose numerical approximation of X(t), following section 1, we take $au_t(k) = (t-|t/T|) + k\Delta t$ for which $d au_t(k) = \Delta t$. Also note that since h(t) is truncated, the maximum range of the samples will correspond to $20/\Delta t$ and will only be non-zero for k>0. Thus, we get,

$$Y(t)pprox \sum_{k=0}^{20/\Delta t} X(\lfloor t/T
floor - k\Delta t) e^{-a((t-\lfloor t/T
floor)+k\Delta t)} \Delta t.$$

We implement the process Y(t) in the code below and plot it for different values of the parameter a.

```
Y_t (generic function with 1 method)
 1 begin
 2
       # define some useful parameters
 3
       alpha1 = 0;
       alpha2 = 0.2;
 4
 5
 6
       # define some related functions
 7
       t_t(t, k) = (t - k_t(t)) + k * deltaT;
       Y(t, X_k, a, T, m) = sum([X_t(t - t_t(t, k), X_k, T, m) * exp(-a * t_t(t, k)))
 8
       for k = 0:(20/deltaT)]) * deltaT;
 9
       Y_t(t, a) = Y(t, Xk, a, T, m);
10 end
MethodError: no method matching floor(::Type{Int64}, ::Vector{Float64})
Closest candidates are:
floor(::Type{T}, !Matched::Missing) where T at missing.jl:154
floor(::Type{T}, !Matched::Rational) where T at rational.jl:458
floor(::Type{T}, !Matched::BigFloat) where T<:Union{Signed, Unsigned} at mpfr.jl:318
  1. k_t(::Vector{Float64}) @ [ Other: 14
  3. X_t(::Float64, ::Vector{Float64}, ::Int64, ::Int64) @ [ Other: 18
  4. (::Main.var"workspace#45".var"#1#2"{Float64, Vector{Float64}, Int64, Int64, Int64})
     (::Float64) @ none:0
  5. iterate @ generator.jl:47 [inlined]
  6. collect(::Base.Generator{StepRangeLen{Float64, Base.TwicePrecision{Float64},
     Base.TwicePrecision{Float64}, Int64}, Main.var"workspace#45".var"#1#2"{Float64,
     Vector{Float64}, Int64, Int64, Int64}}) @ array.jl:787
  7. Y(::Float64, ::Vector{Float64}, ::Int64, ::Int64, ::Int64) @ | Other: 8
  8. Y_t(::Float64, ::Int64) @ | Other: 9
  9. Yt<sub>1</sub>(::Float64) @ | Local: 2
 10. _broadcast_getindex_evalf @ broadcast.jl:670 [inlined]
 11. _broadcast_getindex @ broadcast.jl:643 [inlined]
 12. getindex @ broadcast.jl:597 [inlined]
 13. copy @ broadcast.jl:899 [inlined]
 14. materialize(::Base.Broadcast.Broadcasted{Base.Broadcast.DefaultArrayStyle{1},
     Nothing, typeof(Main.var"workspace#45".Yt1), Tuple{StepRangeLen{Float64,
     Base.TwicePrecision{Float64}, Base.TwicePrecision{Float64},
     Int64}}}) @ broadcast.jl:860
 15. top-level scope @ | Local: 3
 1 begin
 2
       Yt_1(t) = Y_t(t, alpha1);
       plot(\underline{ts}, Yt_1.(\underline{ts}); legend=false)
 3
       xlabel!("Time [s]");
 4
 5
       ylabel!("Amplitude");
```

title!(L"Y(t; a=0)");

plot!(; xlims=(ts[1],ts[end]))

6 7

8 end

3. Power Spectral Density of Random Processes

For LTI systems, we have the following statistical result,

$$S_Y(f) = S_X(f)|H(f)|^2,$$

where S(f) is the power spectral density (PSD) of the respective signals and H(f) is the Fourier transform of the impulse response. Recall that, in section 1, we have treated X_k as samples of X(t) every T second intervals. For a bandlimited WGN process we have the result that the samples $X_k \sim \mathcal{N}(0, N_0 B)$ where $N_0 = 2$ is the net power of the signal X(t) and $B = \frac{1}{2}$ is its bandwidth. Therefore, we have $S_X(f) = 1 = \frac{N_0}{2}$ for $|f| < \frac{1}{2}$. Furthermore, we have $H(f) = \frac{1 - e^{-20(a + j2\pi f)}}{a + j2\pi f}$, and to get the squared magnitude of H, we multiply it with its conjugate H^* . Finally, we have,

$$|H(f)|^2 = rac{1 + e^{-20a}(e^2 - 2\cos(40\pi f))}{a^2 + 4\pi^2 f^2}$$

We expect $S_Y(f)=\frac{N_0}{2}|H(f)|^2=|H(f)|^2$, for $|f|<\frac{1}{2}$ and zero elsewhere. Thus, the theoretical PSDs of Y(t) for filters with different values of parameter a are,

$$S_Y(f) = \left\{ egin{array}{ll} rac{1 + (e^2 - 2\cos(40\pi f))}{4\pi^2 f^2}, & a = 0 \ \ rac{1 + e^{-4}(e^2 - 2\cos(40\pi f))}{0.04 + 4\pi^2 f^2}, & a = 0.2 \end{array}
ight.$$

Note that for for a=0, S_Y is unbounded near f=0, whereas for a=0.2 it only reaches a maximum of pprox 27.46 at 0 Hz. We have created a function to get the theoretical PSD for Y(t) below.