plot_sampled_function(gammatone, fs=1000, tlim=(0, 0.1), tscale=1e3, tunits="msecs", f=100, n=4, phi=0.0, a=1.0) Sampled Function 1.00 0.75 0.50 0.25 Amplitude 0.00 -0.25-0.50-0.75-1.001000 0 2000 3000 4000 5000 6000 Time (msecs) Sampled Function 1e-7 Continuous 1.0 0.5 Amplitude 0.0 -0.5-1.020 40 60 80 100 Time (msecs) Niquist Frequency acts as a maximum before aliasing $f_N=rac{f_s}{2}$ Aliasing occurs when the frequency of the signal is greater than the Nyquist frequncy $f>f_N$ In [3]: # Below Nyquist: fs = 10 Hz, f = 2 Hz (clearly sampled with only a few points per cycle) plot_sampled_function(sinewave, fs=10, tlim=(0, 2*np.pi), tscale=1e3, tunits="msecs", f=2) # At Nyquist: fs = 10 Hz, f = 5 Hz plot_sampled_function(sinewave, fs=10, tlim=(0, 2*np.pi), tscale=1e3, tunits="msecs", f=5) # Cosine at Nyquist: using cosine via phase shift plot_sampled_function(lambda t, **kw: np.cos(2*np.pi*5*t), fs=10, tlim=(0, 2*np.pi), tscale=1e3, tunits="msecs") # Above Nyquist: fs = 10 Hz, f = 7.5 Hz, which aliases plot_sampled_function(lambda t, **kw: np.cos(2*np.pi*7.5*t), fs=10, tlim=(0, 2*np.pi), tscale=1e3, tunits="msecs") Sampled Function 1.00 0.75 0.50 0.25 Amplitude 0.00 -0.25-0.50-0.75-1.001000 2000 3000 4000 5000 6000 Time (msecs) Sampled Function 1.00 0.75 0.50 0.25 Amplitude 0.00 -0.25-0.50-0.75-1.001000 2000 3000 4000 5000 6000 0 Time (msecs) Sampled Function 1.00 0.75 0.50 0.25 Amplitude 0.00 -0.25-0.50-0.75-1.001000 2000 3000 4000 5000 6000 Time (msecs) Sampled Function 1.00 0.75 0.50 0.25 Amplitude 0.00 -0.25-0.50-0.75-1.001000 2000 3000 4000 5000 6000 Time (msecs) Dirac Delta Function Used to model an impulse or discrete event as a brief impulse of energy. $\delta(t) = \left\{ egin{array}{ll} ext{undefined} & t=0, \ 0 & t
eq 0. \end{array}
ight.$ but also, $\int_{-\infty}^{\infty} \delta(t) dt = 1$ another way to interpret it is, $\int_{-\infty}^{\infty} f(t)\delta(t- au)dt = f(au)$ ullet where $\delta(t- au)$ is zero everywhere except at t= auullet and at the infinitesimal point f(au) is a constant and so so multiplies the integral, which is one. Unit Step Function The step function is used to indicate a constant signal that starts at t=0. $u(t) = egin{cases} 1 & t \geq 0, \ 0 & t < 0. \end{cases}$ In [4]: # Example plots: t = np.linspace(-0.01, 0.01, 200)plt.figure(figsize=(8,3)) plt.plot(t, delta(t, fs=1000), label="delta(t)") plt.plot(t, u(t), label="u(t)") plt.xlabel("Time (s)") plt.legend() plt.grid(True) plt.show() delta(t) u(t) 0.8 0.6 0.4 0.2 0.0 -0.0100 -0.0075 -0.0050 -0.0025 0.0000 0.0025 0.0050 0.0075 0.0100 Time (s) In [5]: # Example usage: a gammatone signal delayed by 0.025 s and lasting 0.1 s t = np.linspace(0, 0.2, 1000)x = gensignal(t, gammatone, tau=0.025, T=0.1, f=100, n=4, phi=0.0, a=1.0)plt.figure(figsize=(8,3)) plt.plot(t*1e3, x) plt.xlabel("Time (msecs)") plt.title("Gammatone signal via gensignal") plt.grid(True) plt.show() Gammatone signal via gensignal 1.0 0.5 0.0 -0.5-1.025 50 75 100 125 150 175 200 Time (msecs) Power the average energy over a period. $E_x = \sum_{}^{N} \left| x[n]
ight|^2$ n=1power of x is then, $P_x = rac{1}{N}\sum_{n=1}^N \left|x[n]
ight|^2 = \sigma_x^2$ for a signal with additive noise $y[t] = x[t] + \epsilon[t]$ the SNR is simply However, it is usually described in decibals $dB~SNR = 10 log_{10}(rac{P_x}{P_\epsilon}) = 20 log_{10}(rac{\sigma_x}{\sigma_\epsilon})$ Peak Signal Noise Ratio Is used more often in image processing. $PSNR = 10log_{10}(rac{max_{t}(y[t])^{2}}{\sigma_{y}^{2}}) = 20log_{10}(rac{max_{t}(y[t])}{\sigma_{y}})$ In [6]: # Example: Noisy sinewave signal. t = np.linspace(0, 2, 1000)y = noisysignal(t, sinewave, tau=0.5, T=1, sigma=0.2, f=5, d=0.0)plt.figure(figsize=(8,3)) plt.plot(t, y, label="Noisy Signal") plt.xlabel("Time (s)") plt.legend() plt.grid(True) plt.show() 1.5 Noisy Signal 1.0 0.5 0.0

-0.5

-1.0

-1.5

In [7]: # Example:

In [8]: # Demonstration:

In [9]: import numpy as np

0.00

 $desired_dBsnr = 10$

if ind is not None:

Save to a .wav file.

0.25

t = np.linspace(0, 2, 1000)

ind = extent(y_test, theta=0.05)

Ps = power(x[sig_range])

Estimated signal extent: (0, 999)
Estimated SNR (dB): 8.411478733825959

from scipy.io.wavfile import write

print("Estimated signal extent:", ind)

sig_range = slice(ind[0], ind[1]+1)

Pn = power(y_test[sig_range] - x[sig_range])
print("Estimated SNR (dB):", snr(Ps, Pn))

Synthesize a 5-second waveform with 20 random gammatones.

0.50

Suppose x is our signal generated over a time vector.

x = gensignal(t, sinewave, tau=0.5, T=1, f=5, d=0.0)

sigma_required = snr2sigma(x, dBsnr=desired_dBsnr)

Noise sigma for 10 dB SNR: 0.15803476716592346

0.75

print("Noise sigma for", desired_dBsnr, "dB SNR:", sigma_required)

 $y_test = noisysignal(t, sinewave, tau=0.5, T=1, sigma=0.2, f=5, d=0.0)$

One could then compute the SNR using the signal's energy over that extent

t, waveform = grand_synthesis(duration=5, fs=44100, num_tones=20, T=0.1, fmin=200, fmax=2000)

1.00

Time (s)

1.25

1.50

1.75

2.00

In [1]: from A3a_rhl72 import *

In [2]: plot_sampled_function(sinewave, fs=5, tlim=(0, 2*np.pi), tscale=1e3, tunits="msecs", f=1.0, d=0.0)

write("synthesized.wav", 44100, (waveform * 32767).astype(np.int16))
print("Synthesized audio saved as synthesized.wav")

Synthesized audio saved as synthesized.wav