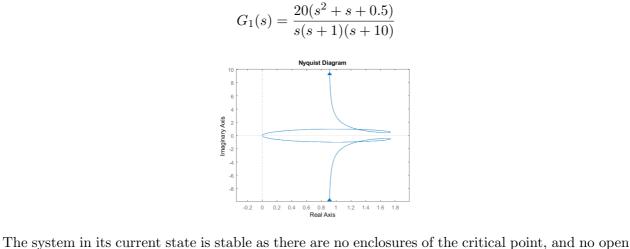
Daniel Eisen: 300447549

July 31, 2021

1. (a)

Section A - Formative Questions



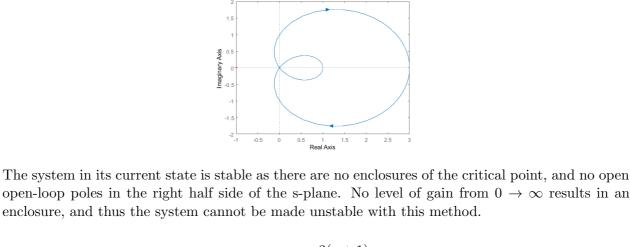
enclosure, and thus the system cannot be made unstable with this method. $G_2(s) = \frac{20(s^2 + s + 0.5)}{s(s-1)(s+10)}$

open-loop poles in the right half side of the s-plane. No level of gain from $0 \to \infty$ results in an

(b)

(c) $G_3(s) = \frac{s^2 + 3}{(s+1)^2}$

will be no enclosure of the critical point and the system can be driven unstable.



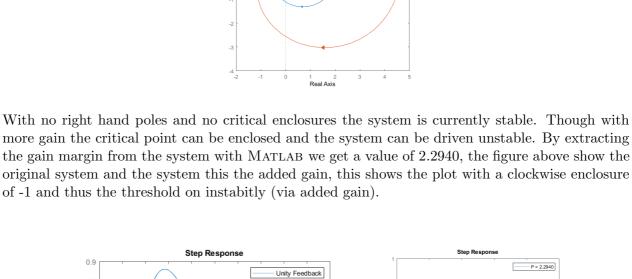
 $G_4(s) = \frac{3(s+1)}{s(s-10)}$

(d)

 $G = e^{-0.2s} \frac{4}{s+2}$ Eq: Delayed transfer function

2.

(a)

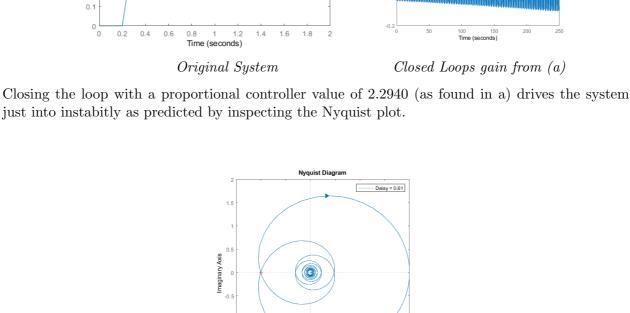


0.3 0.2

(b)

0.8

enclosure divergence angle.

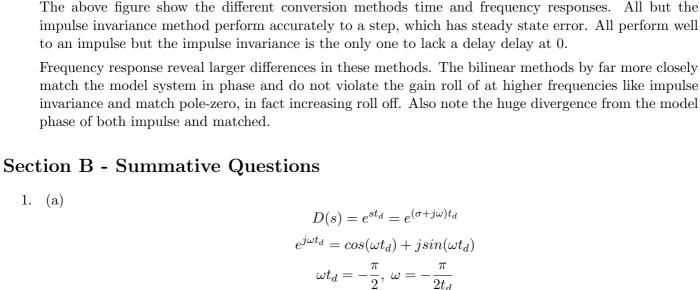


The figure above shows that a delay time of 0.61 causes the Nyquist plot to enclose the critical point. Though note a differences from the increased gain plot, smaller overall plot and a tighter

1. (a)

(b)

3.



 $\omega = 1000 rad/s, \ \phi = 15^{\circ} = \frac{\pi}{12}$

 $t_d = \frac{\pi}{1200} \approx 0.00261799388$

Figure above show the resultant gain of the frequency response of these approximations. This results in a ts 0.00027 for the 1st and 0.000261 for the 2nd. Both already close to the model delay.

 $sampler = \frac{1 - e^{st_s}}{s}$

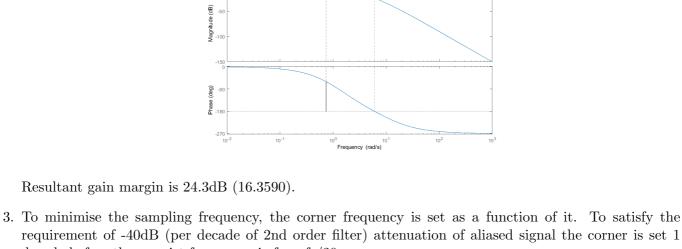
2nd Pade

(c) Using Matlab to model the 1st and 2nd Pade approximations and modifying ts until $\phi = 15^{\circ}$.

 $pade: e^{st_s} \approx \frac{1 - s\frac{t_2}{2}}{1 + s\frac{t_s}{2}}$ $substitute \rightarrow sampler \approx \frac{2}{s + \frac{s}{t_a}}$

1st Pade

Sampler inserts a left hand pole at 2/sampling time. Therefore as the sampling speed increases, the inserted pole becomes less and less dominant, the gain margin increases and the system more accurately represents the unsampled system. Figures above shows this progression with (left to right) low sampling speed, higher sampling speed and unsampled system. For 10x: Unity gain frequency is 3.55rad/s = 0.565Hz, new sampling: f = 5.56Hz, $t_s = 0.177$

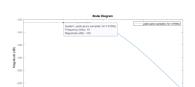


decade before the nyquist frequency, ie fc = fs/20. At minimum, fs must be set to 100kHz (fc = 5kHz), but this exceeds the phase margin of the system (10 degrees.)

In Matlab the sampling frequency is increased (retaining the fc relationship), and computing the phase at 10kHz of a sampled Butterworth filter until it reduced to 10 degrees.

fs = 1.81Mhz, fc = fs/20

Though note the gain of the sampled system is heavily reduced so this would still need to the scaled.



Frequency Response (using 1st Pade delay)