Control Theory (00350188) lecture no. 12

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Outline

Analog redesign

Discrete-time design

Discretized plant and its properties

Classical methods for discrete systems (mostly stability)

Outline

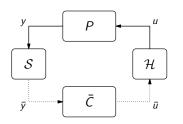
Analog redesign

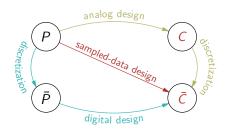
Discrete-time design

Discretized plant and its properties

Classical methods for discrete systems (mostly stability)

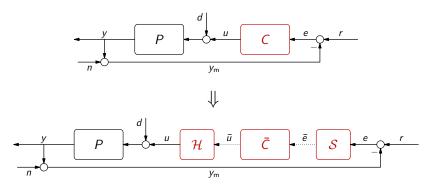
Three approaches to sampled-data control design





- Digital redesign of analog controllers
 (do your favorite analog design first, then discretize the resulting controller)
- Discrete-time design
 (discretize the problem first, then do your favorite discrete design)
- 3. Direct digital (sampled-data) design (design discrete-time controller $\bar{C}(z)$ directly for analog specs

The redesign problem



Starting point:

- "good" analog controller C (designed by whatever method)

Goal:

- find \bar{C} such that $\mathcal{H}\bar{C}\mathcal{S}\approx C$

(we consider $S = S_{idl}$, $\mathcal{H} = \mathcal{H}_{ZOH}$, and periodic sampling with given h > 0).

Discrete transfer functions (from LS)

continuous-time systems

discrete-time systems

differential equations

Laplace transform

s is the derivative in the time domain

left-half plane in the *s*-plane : —

imaginary axis, $s=\mathrm{j}\omega$

static gain is $G(s)|_{s=0} = G(0)$

integral action: pole at s=0

difference equations

z transform

z is the shift in the time domain



-: unit disk in the z-plane

unit circle, $z = e^{j\theta}$

static gain is $G(z)|_{z=1} = G(1)$

integral action: pole at z = 1

Choice of \bar{C} : ad hoc approaches

Logic: imitate C, often based on approximating derivatives, like

fwd Euler:
$$\dot{x}(ih) \approx \frac{x(ih+h) - x(ih)}{h}$$
 $\Longrightarrow s \to \frac{z-1}{h}$

bwd Euler: $\dot{x}(ih) \approx \frac{x(ih) - x(ih-h)}{h}$ $\Longrightarrow s \to \frac{z-1}{hz}$

Tustin¹: $\frac{\dot{x}(ih+h) + \dot{x}(ih)}{2} \approx \frac{x(ih+h) - x(ih)}{h}$ $\Longrightarrow s \to \frac{2z-1}{hz+1}$

making sense if h is "small enough."

¹Matlab: c2d(C,h,'tustin'), where C is a continuous-time system.

Choice of \bar{C} : ad hoc approaches

Logic: imitate C, often based on approximating derivatives, like

fwd Euler:
$$\dot{x}(ih) \approx \frac{x(ih+h)-x(ih)}{h}$$
 $\Longrightarrow s \to \frac{z-1}{h}$ bwd Euler: $\dot{x}(ih) \approx \frac{x(ih)-x(ih-h)}{h}$ $\Longrightarrow s \to \frac{z-1}{hz}$

Tustin:
$$\frac{\dot{x}(ih+h)+\dot{x}(ih)}{2}\approx\frac{x(ih+h)-x(ih)}{h}\implies s\to \frac{2}{h}\frac{z-1}{z+1}$$

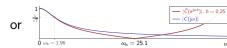
making sense if h is "small enough."

Example: If C(s) = 2/(s+2), then

$$|\bar{C}(z)| = C(s)|_{s=\frac{2}{h}\frac{z-1}{z+1}} = \frac{2}{2/h\cdot(z-1)/(z+1)+2} = \frac{h(z+1)}{(h+1)z+h-1}.$$

and

$$|ar{\mathcal{C}}(\mathrm{e}^{\mathrm{j}\omega h})| = \frac{1}{\sqrt{2}} \int_{0}^{1} \frac{|\mathcal{C}(\mathrm{e}^{\mathrm{j}\omega h})|, h = 0.6}{|\mathcal{C}(\mathrm{g}\omega)|}}{\int_{0}^{1} \frac{|\mathcal{C}(\mathrm{g}\omega)|}{|\mathcal{C}(\mathrm{g}\omega)|}} \quad \text{of} \quad$$



General bilinear transformation

Given $\gamma > 0$, consider the mapping (Tustin corresponds to $\gamma = 2/h$)

$$s \to \gamma \, \frac{z-1}{z+1} \iff z \to \frac{\gamma+s}{\gamma-s}$$

between s and z complex planes. Every $s = \sigma + j\omega$ is mapped to

$$z = \frac{\gamma + (\sigma + j\omega)}{\gamma - (\sigma + j\omega)} \implies |z|^2 = \frac{(\gamma + \sigma)^2 + \omega^2}{(\gamma - \sigma)^2 + \omega^2}.$$

Hence,

$$|z|^2 - 1 = \frac{2\gamma\sigma}{(\gamma - \sigma)^2 + \omega^2}$$

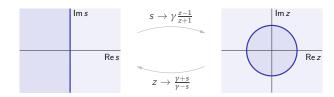
and we end up with the relations

$$-|z|<1\iff \sigma=\operatorname{Re} s<0$$

$$-|z| > 1 \iff \sigma = \operatorname{Re} s > 0$$

$$-|z|=1\iff \sigma=\operatorname{Re} s=0$$

General bilinear transformation (contd)



Thus,

- any "stable" s is mapped to a "stable" z
- any "unstable" s is mapped to a "unstable" z
- any "borderline" s is mapped to a "borderline" z

Moreover.

- any CT frequency ω is mapped to the DT frequency $\theta = 2 \arctan(\omega/\gamma)$ (i.e. bilinear transformations squeeze the whole \mathbb{R} to \mathbb{T} , with no folding effects)
- the lowest $\omega = 0$ is mapped to the lowest $\theta = 0$
- the highest $\omega = \pm \infty$ is mapped to the highest $\theta = \pm \pi$

Choice of \bar{C} : more about Tustin

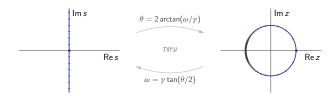
lf

$$\bar{C}(z) = C(s)|_{s=\frac{2}{h}\frac{z-1}{z+1}},$$

then

- \bar{C} is stable iff C is stable,
- $ar{\mathcal{C}}$ is unstable iff \mathcal{C} is unstable,
- the number of integrators in $\bar{C}(z)$ equals that in C(s),
- $\bar{C}(z)$ is bi-proper, unless C(s) has either poles or zeros at s=2/h
 - zero multiplicity of C(s) at s=2/h equals the pole excess of $\bar{C}(z)$
 - pole multiplicity of C(s) at s=2/h equals the zero excess of $\bar{C}(z)$
- $\bar{C}(z)$ has a zero at z=-1 of multiplicity m iff C(s) is strictly proper and its pole excess is m
- $\bar{C}(z)$ has a pole at z=-1 of multiplicity m iff C(s) is non-proper and its zero excess is m

Discretizing C: Tustin with pre-warping



The nonlinear warping of the frequency mapping is not ideal. We would be happier with $\theta = \omega h$ in low frequencies (if folding effects are insignificant). This happens only at $\omega = 0$ and the frequency $\omega_{\text{nowarp}} \in (0, \omega_{\text{N}})$, at which

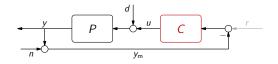
$$2\arctan\frac{\omega_{\mathsf{nowarp}}}{\gamma} = \omega_{\mathsf{nowarp}} h \;\iff\; \gamma = \omega_{\mathsf{nowarp}}\cot\frac{\omega_{\mathsf{nowarp}}h}{2} \in \left(0,\frac{2}{h}\right).$$

The bilinear transformation with γ as above for a given $\omega_{\text{nowarp}} \in (0, \omega_{\text{N}})$ is known¹ as Tustin with pre-warping. As $\omega_{
m nowarp} o 0$, the ordinary Tustin for $\gamma = 2/h$ is recovered, for which $d\bar{C}(e^{j\theta})/d\theta|_{\theta=0} = dC(\omega)/d\omega|_{\omega=0}$ as well.

¹Matlab: c2d(C,h,c2d0ptions('Method','tustin','PrewarpFrequency',w0)).

Example: DC motor

A DC motor from Lecture 1, controlled in closed loop



Requirements:

- closed-loop stability (of course)
- zero steady-state error for a step in r
- zero steady-state error for a step in d integrator in C(s)

always holds

- good stability margins
- $-\omega_{\rm c}\approx 2\,[{\rm rad/sec}]$

Design:

LQG loop shaping, with a PI weight W (like in Lecture 11)

Example: analog design

Weight:

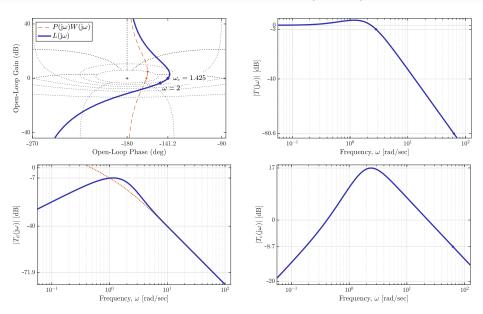
$$W(s) = 5.06\left(1 + \frac{1}{s}\right)$$

Controller:

$$C(s) = W(s)C_{a}(s) = \frac{23.081(s + 2.075)(s + 0.5346)}{s(s^{2} + 6.155s + 17.44)}$$

(a pole of $C_{\rm a}(s)$ cancels the zero of W(s) at s=1). The actual crossover is $\omega_{\rm c}=1.4248$ and the closed-loop bandwidth is $\omega_{\rm b}=2.8155$.

Example: analog design (contd)



Example: controller discretization

Using Tustin, the discretized controllers are

$$h = 0.01: \ \bar{C}(z) = \frac{0.11337(z+1)(z-0.9795)(z-0.9947)}{(z-1)(z^2-1.939z+0.9403)}$$

$$h = 0.1: \ \bar{C}(z) = \frac{0.96777(z+1)(z-0.812)(z-0.9479)}{(z-1)(z^2-1.415z+0.5445)}$$

$$h = 0.5: \ \bar{C}(z) = \frac{2.7375(z+1)(z-0.317)(z-0.7642)}{(z-1)(z^2+0.04973z+0.152)}$$

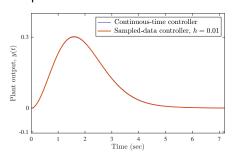
All of them

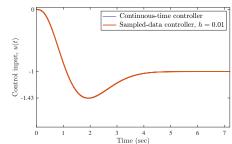
- preserve integral actions (pole at $s = 0 \rightarrow \text{pole at } z = 1$)
- have single zeros at z = -1 (pole excess of P(s) is 1)
- are bi-proper

which are general properties of the Tustin transformation.

Example: $d(t) = \mathbb{1}(t)$ and n(t) = 0

Responses with h = 0.01:





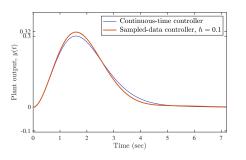
sampled-data response \approx analog response

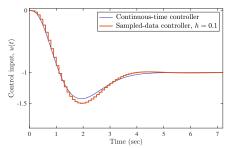


adequate sampling rate

Example: d(t) = 1(t) and n(t) = 0

The same with h = 0.1:





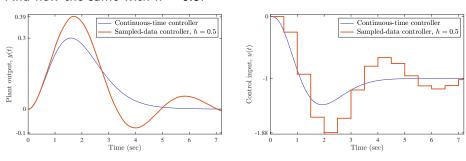
sampled-data response starts getting worse than analog response



sampling rate starts to become problematic

Example: $d(t) = \mathbb{1}(t)$ and n(t) = 0

And now the same with h = 0.5:



sampled-data response is substantially worse than analog response

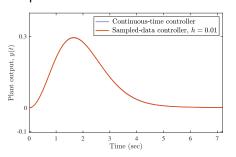


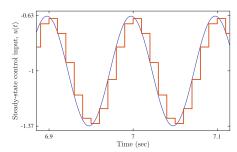
sampling rate becomes inadequate

(further increase of h eventually results in an unstable closed-loop system).

Example: $d(t) = \mathbb{I}(t)$ and $n(t) = \sin(20\pi t + 0.1)$

Responses with h = 0.01:





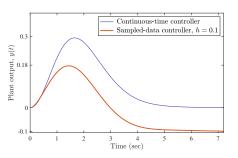
sampled-data response \approx analog response

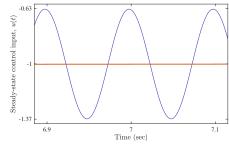


adequate sampling rate

Example: $d(t) = \mathbb{I}(t)$ and $n(t) = \sin(20\pi t + 0.1)$

Now the same with h = 0.1:





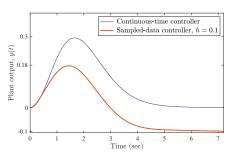
Oops,

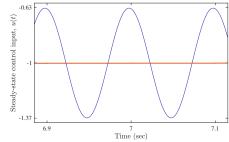
sampled-data response is qualitatively different from analog response



Example: $d(t) = \mathbb{I}(t)$ and $n(t) = \sin(20\pi t + 0.1)$

Now the same with h = 0.1:





Oops,

 sampled-data response is qualitatively different from analog response (steady-state error is nonzero, the harmonic of measurement noise disappears)

Why?

The sampled-data controller

$$\frac{\int_{0}^{T} \int_{0}^{T} \int_$$

We now know that in the frequency domain

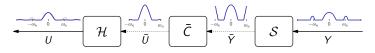
- S causes aliasing by folding ultra- ω_N frequencies of $Y(j\omega)$ to $[-\omega_N, \omega_N]$ of $\bar{Y}(e^{j\omega h})$
- \bar{C} acts as a standard LTI filter, $\bar{U}(e^{j\omega h}) = \bar{C}(e^{j\omega h})\bar{Y}(e^{j\omega h})$
- ${\cal H}$ clones $[-\omega_{\scriptscriptstyle {
 m N}},\omega_{\scriptscriptstyle {
 m N}}]$ frequency interval of $\bar U({\rm e}^{{
 m j}\omega h})$ to all ${\mathbb R}$ and filters the result by the low-pass hF_{ϕ} , where $F_{\phi}(j\omega) = (1 - e^{-j\omega h})/(j\omega h)$

In other words.

$$U(\mathrm{j}\omega) = \frac{1 - \mathrm{e}^{-\mathrm{j}\omega h}}{\mathrm{j}\omega h} \, \bar{C}(\mathrm{e}^{\mathrm{j}\omega h}) \sum_{i \in \mathbb{Z}} Y(\mathrm{j}(\omega + 2\omega_{\mathrm{N}}i))$$

Effects of aliasing

If aliased parts remain qualitatively unchanged, then aliasing is harmless

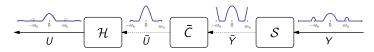


But if they migrate to different frequency bands, then the picture changes

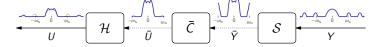
(red dotted lines correspond to the spectrum of Cy).

Effects of aliasing

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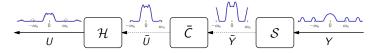


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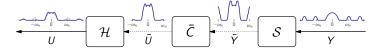
Moral



Once high-frequency components of y alias as low-frequency ones and blend with low-frequency components of y,

- nothing can be done via a "better" processing by $ar{\mathcal{C}}(z)$.

Moral



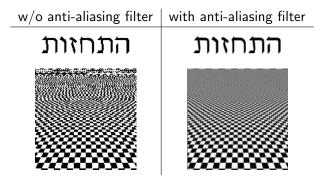
Once high-frequency components of y alias as low-frequency ones and blend with low-frequency components of y,

- nothing can be done via a "better" processing by C(z).

The only way to cope with this phenomenon is to

- filter out those frequencies in continuous time, before sampling (nip them in the bud). Low-pass filters doing that are known as
 - anti-aliasing filters.

Anti-aliasing filtering: non-control examples



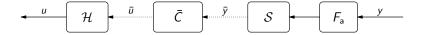
where anti-aliasing filters used are

- noncausal low-pass filters with the bandwidth $\omega_{ extsf{N}}.$

Best choice, performance-wise, is

- the ideal low-pass filter F_{ilp} with the bandwidth $\omega_{\text{b}} = \omega_{\text{N}}$, whose impulse response $f_{\text{ilp}}(t) = \text{sinc}(\omega_{\text{N}}t)$, but it is hard to implement.

Anti-aliasing filters in feedback loops



Additional considerations:

- must be causal,
- $\ | extstyle F_{\mathsf{a}}(\mathsf{j}\omega) | \ll 1$ for all $\omega \geq \omega_{\mathsf{N}}$,
- avoid adding a substantial phase lag around the crossover.

introduce phase large before the magnitude starts to decay

Hence

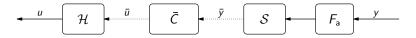
the bandwidth $\omega_{
m b}$ of $F_{
m a}$ should be well below $\omega_{
m N}$

and, as a result

the choice of the Nyquist frequency should be conservative

(conventional wisdom has it that $\omega_N \geq 10 \div 30\,\omega_c$, where ω_c is the analog crossover

Anti-aliasing filters in feedback loops



Additional considerations:

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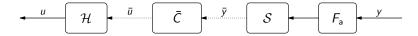
We already know (Lecture 1) that finite-dimensional low-pass filters

introduce phase lags before the magnitude starts to decay.

Hence,

 $-\,$ the bandwidth ω_{b} of F_{a} should be well below ω_{N}

Anti-aliasing filters in feedback loops



Additional considerations:

- must be causal,
- $-|F_{\mathsf{a}}(\mathsf{j}\omega)|\ll 1$ for all $\omega\geq\omega_{\mathsf{N}}$,
- avoid adding a substantial phase lag around the crossover.

We already know (Lecture 1) that finite-dimensional low-pass filters

introduce phase lags before the magnitude starts to decay.

Hence,

- the bandwidth ω_{b} of F_{a} should be well below ω_{N}
- and, as a result
 - the choice of the Nyquist frequency should be conservative (conventional wisdom has it that $\omega_{\rm N} \geq 10 \div 30 \,\omega_{\rm c}$, where $\omega_{\rm c}$ is the analog crossover)

Aliasing: example (contd)

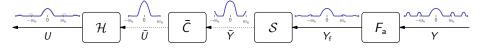
Let

$$F_{\mathsf{a}}(s) = rac{\omega_{\mathsf{b}}^2}{s^2 + \sqrt{2}\omega_{\mathsf{b}}s + \omega_{\mathsf{b}}^2}, \quad \omega_{\mathsf{b}} = rac{\omega_{\mathsf{N}}}{5} = 0.4\pi$$

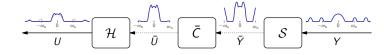
(second-order Butterworth with $|F_a(j\omega)| = 1/\sqrt{1+(\omega/\omega_b)^4}$). In this case



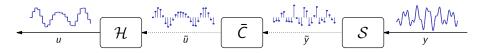
and



Compare with



Effects of ${\cal H}$



With

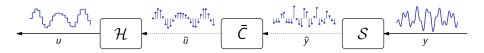
$$U(j\omega) = h \frac{1 - e^{-j\omega h}}{j\omega h} \, \bar{U}(e^{j\omega h})$$

we effectively (factor h is offset by 1/h of the sampler) have the FIR

$$F_{\phi}(s) = \frac{1 - e^{-sh}}{sh}$$

in the loop.

Effects of \mathcal{H}



With

$$U(j\omega) = h \frac{1 - e^{-j\omega h}}{j\omega h} \, \bar{U}(e^{j\omega h})$$

we effectively (factor h is offset by 1/h of the sampler) have the FIR

$$F_{\phi}(s) = \frac{1 - e^{-sh}}{sh}$$

in the loop. This is a low-pass filter, whose frequency response

$$F_{\phi}(j\omega) = \operatorname{sinc} \frac{\omega h}{2} e^{-\omega h/2} =$$

This is a low-pass F_{ϕ} , having a phase lag $\omega h/2$ for $0 \le \omega \le \omega_N$.

Effects on the design of C

Things to remember:

- sample sufficiently fast
- account for an anti-aliasing filter F_{a}
- account for low-pass in the ZOH, F_{ϕ}

unless sensor is digital

unless sensor is digital

Effects on the design of C

Things to remember:

- sample sufficiently fast
- account for an anti-aliasing filter F_a
- account for low-pass in the ZOH, F_{ϕ}

Advised to

- design C for the augmented $F_a P F_{\phi}$.

Remark Because $F_{\phi}(s)=(1-\mathrm{e}^{-sh})/(sh)$ is infinite dimensional, its approximations

$$F_{\phi}(s)pprox F_{\phi, ext{del}}:=\mathrm{e}^{-sh/2} \quad ext{or} \quad F_{\phi}(s)pprox F_{\phi,2}:=rac{12}{h^2s^2+6hs+12}$$

can be used in analytic design methods, such as state space. Note that $F_{\phi,2}(s)$ is the [1,2]-Padé approximant of $F_{\phi}(s)$, whose bandwidth $\omega_{\rm b}\approx 2.7233/h<\omega_{\rm N}$ and frequency response in $\omega\in[0,\omega_{\rm N}]$ is not far from that of F_{ϕ} , e.g.

$$F_{\phi,2}(\mathrm{j}\omega_{\mathrm{N}}) pprox 0.6326 \mathrm{e}^{-\mathrm{j}1.4583} \quad \mathrm{vs.} \quad F_{\phi}(\mathrm{j}\omega_{\mathrm{N}}) = rac{2}{\pi} \mathrm{e}^{-\mathrm{j}\pi/2} pprox 0.6366 \mathrm{e}^{-\mathrm{j}1.5708}.$$

Outline

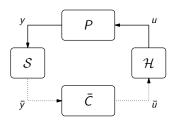
Analog redesign

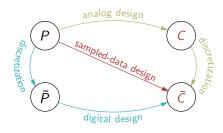
Discrete-time design

Discretized plant and its properties

Classical methods for discrete systems (mostly stability)

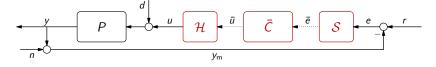
Three approaches to sampled-data control design





- 1. Digital redesign of analog controllers
- 2. Discrete-time design (discretize the problem first, then do your favorite discrete design)
- 3. Direct digital (sampled-data) design

What does \bar{C} see?

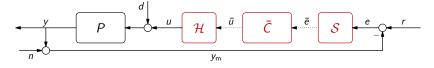


Input:

$$\bar{e} = Sr - Sn - SPd - SPH\bar{u},$$

where \bar{u} is its output (cf. the analog e = r - n - Pd - Pu).

What does \bar{C} see?



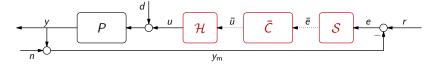
Input:

$$\bar{e} = Sr - Sn - SPd - SPH\bar{u},$$

where \bar{u} is its output (cf. the analog e=r-n-Pd-Pu). Observations:

- the discrete $ar{P}_h := \mathcal{S}\mathcal{P}\mathcal{H}: ar{u} \mapsto ar{y}$ is the plant from the viewpoint of $ar{\mathcal{C}}$,
- sampled reference signal $\bar{r} := Sr$ replaces r,
- sampled noise signal $\bar{n} := Sn$ replaces n,
- ${\cal S}Pd$ doesn't fit, unless we assume that $dpprox {\cal H}ar d$ for some ar d

What does \bar{C} see?



Input:

$$\bar{e} = Sr - Sn - SPd - SPH\bar{u}$$
,

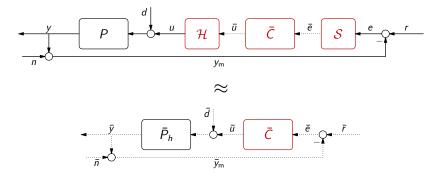
where \bar{u} is its output (cf. the analog e = r - n - Pd - Pu). Observations:

- the discrete $\bar{P}_h := SP\mathcal{H} : \bar{u} \mapsto \bar{y}$ is the plant from the viewpoint of \bar{C} ,
- sampled reference signal $\bar{r} := Sr$ replaces r,
- sampled noise signal $\bar{n} := Sn$ replaces n,
- SPd doesn't fit, unless we assume that $d \approx \mathcal{H} \bar{d}$ for some \bar{d} In other words,

$$ar{e} pprox ar{r} - ar{n} - \mathcal{S}P\mathcal{H}ar{d} - \mathcal{S}P\mathcal{H}ar{u} = ar{r} - ar{n} - ar{P}_har{d} - ar{P}_har{u}$$

(if d can be viewed as piecewise constant, like u, in the case of $\mathcal{H} = \mathcal{H}_{ZOH}$).

What does \bar{C} see? (contd)



meaning that \bar{C} lives in a pure discrete (stroboscopic) world and this world approximates the reality well if

- disturbance d may be approximated by a piecewise-constant $\mathcal{H} \bar{d}$ it is still assumed that $\mathcal{H} = \mathcal{H}_{\sf ZOH}$
- sampler $S = S_{idl}F_a$ and F_a filters out ultra-Nyquist frequencies of n or the sensor is digital, like an encoder, in which case noise is digital by nature

Outline

Analog redesign

Discrete-time design

Discretized plant and its properties

Classical methods for discrete systems (mostly stability)

Discretization

Our task is to find

$$ar{P}_{h} = \mathcal{S}\mathcal{P}\mathcal{H} = \mathcal{S}_{\mathsf{idl}} \overbrace{F_{\mathsf{a}}\mathcal{P}}^{P_{\mathsf{a}}} \mathcal{H}_{\mathsf{ZOH}}$$

for given LTI P and F_a . Let

$$P_{a}:\begin{cases} \dot{x}(t) = Ax(t) + Bu(t), & x(0) = 0\\ y(t) = Cx(t) \end{cases}$$

 $(P_a(s))$ is always strictly proper, for so is $F_a(s)$). Because

- having ${\cal H}$ at the input implies that $u={\cal H}_{ extsf{ZOH}}ar{u}$ for some discrete $ar{u}$,
- having \mathcal{S}_{idl} at the output implies that only $\bar{y}[i] = y(ih)$ is of interest, finding \bar{P}_h is
 - equivalent to finding the mapping $\bar{u} \mapsto \bar{y}$.

Discretization (contd)

Define $\bar{x}[i] := x(ih)$. For a given $\bar{x}[i]$,

$$\bar{x}[i+1] = e^{Ah}\bar{x}[i] + \int_{ih}^{(i+1)h} e^{A(ih+h-s)}Bu(s)ds$$

Because $u(t) = \bar{u}[i]$ for all $t \in (ih, (i+1)h]$, we have that

$$\bar{x}[i+1] = e^{Ah}\bar{x}[i] + \int_{ih}^{(i+1)h} e^{A(ih+h-s)} ds B\bar{u}[i] = e^{Ah}\bar{x}[i] + \int_{0}^{h} e^{As} ds B\bar{u}[i]$$

Because $\bar{y}[i] = y(ih) = C\bar{x}[i]$, the mapping $\bar{u} \mapsto \bar{y}$ satisfies the relation

$$\bar{P}_h: egin{cases} \bar{x}[t+1] = \bar{A}\bar{x}[t] + \bar{B}\bar{u}[t], & \bar{x}[0] = 0 \\ \bar{y}[t] = C\bar{x}[t] \end{cases}$$

where $\bar{A} := e^{Ah}$ and $\bar{B} := \int_0^h e^{As} ds B$.

Discretization (contd)

The dynamics

$$\bar{P}_h: \begin{cases} \bar{x}[t+1] = \bar{A}\bar{x}[t] + \bar{B}\bar{u}[t], & \bar{x}[0] = 0\\ \bar{y}[t] = C\bar{x}[t] \end{cases}$$

is a standard LTI discrete system in state space. Its transfer function²,

$$\bar{P}_h(z) = C(zI - \bar{A})^{-1}\bar{B}$$

is always strictly proper, for $\bar{P}_h(\infty) = 0$.

Note that

 $-\lambda \in \operatorname{spec}(A) \subset \mathbb{C} \implies \operatorname{e}^{\lambda h} \in \operatorname{spec}(\bar{A}) \subset \mathbb{C}$

 $A \in \operatorname{Spec}(A) \subset C \longrightarrow C \subseteq \operatorname{Spec}(A) \subset C$

 $-\lambda\in\operatorname{\mathsf{spec}}(A)\subset\mathbb{C}\implies\exists\lambda\in\operatorname{\mathsf{spec}}(A)\subset\mathbb{C}$ such that $\mathrm{e}^{\lambda h}=\lambda$.

²Matlab: Ph=c2d(P,h) or [Ad,Bd]=c2d(A,B,h).

Discretization (contd)

The dynamics

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lf

$$P_{\mathsf{a}}(s) = \frac{b}{s+a}$$

then A = -a, B = b, and C = 1, so that

$$ar{A}=\mathrm{e}^{-ah}$$
 and $ar{B}=\int_0^h\mathrm{e}^{-as}\mathrm{d}sb=rac{1-\mathrm{e}^{-ah}}{a}\,b$

(with well defined $\lim_{a\to 0} \bar{B} = hb$). As a result,

$$\bar{P}_h(z) = C(zI - \bar{A})^{-1}\bar{B} = \frac{(1 - e^{-ah})b/a}{z - e^{-ah}}$$

It has

- one pole, at e^{-ah}, and
- no zeros,

similarly to the continuous-time $P_a(s)$.

lf

$$P_{a}(s) = \frac{1}{s(s+1)} = \frac{1}{s} - \frac{1}{s+1}$$

then by the linearity of the discretization procedure

$$\bar{P}_h(z) = \frac{h}{z-1} - \frac{1 - e^{-h}}{z - e^{-h}} = \frac{(h + e^{-h} - 1)z + 1 - (1 + h)e^{-h}}{(z-1)(z - e^{-h})}$$

This transfer function

- has two poles, at $e^{0h} = 1$ and e^{-h} and
- one zero, at $-(1-(1+h)e^{-h})/(h+e^{-h}-1) \in (-1,0)$

While poles are still exponents of those of $P_a(s)$, the zero is an artefact.

lf

$$P_{a}(s) = \frac{2}{s(s+1)(s+2)} = \frac{1}{s} - \frac{2}{s+1} + \frac{1}{s+2}$$

then by the linearity of the discretization procedure

$$\bar{P}_h(z) = \frac{h}{z - 1} - \frac{2(1 - e^{-h})}{z - e^{-h}} + \frac{1 - e^{-2h}}{z - e^{-2h}}$$

$$= \frac{2h - 3 + 4e^{-h} - e^{-2h}}{2} \frac{(z - z_{h,1})(z - z_{h,2})}{(z - 1)(z - e^{-h})(z - e^{-2h})}$$

where

$$\begin{bmatrix} z_{h,1} \\ z_{h,2} \end{bmatrix} = \begin{bmatrix} -2 + \sqrt{3} \\ -1 \\ -2 - \sqrt{3} \end{bmatrix}$$

Poles follow the already familiar pattern, but now we have

- two zeros, one of which is nonminimum-phase for h < 2.2755

Poles and zeros of $\bar{P}_h(z)$

Poles of $\bar{P}_h(z)$ are simple. If $P_a(s)$ has a pole at $s=p_i$, then

$$\bar{P}_h(z)$$
 has a pole at $z=\mathrm{e}^{p_ih}$ $|\mathrm{e}^{p_ih}|<1\ (=1)\iff \mathrm{Re}\,p_i<0\ (=0)$

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Zeros of $\bar{P}_h(z)$ are a mess. We only know that

- the number of finite zeros of $\bar{P}_h(z)$ is n-1 for almost all h>0
- if $P_a(s)$ has m finite zeros (m < n) at $s = z_i$, then as $h \downarrow 0$
 - m zeros of $\bar{P}_h(z)$ approach $e^{z_i h}$,
 - the remaining n-m-1 zeros, aka sampling zeros, approach the roots of Euler–Frobenius polynomials $Q_{n-m-1}(z)$, independent of $P_a(s)$:

n-m	$Q_{n-m-1}(z)$
2	z+1
3	$z^2 + 4z + 1$
4	$z^3 + 11z^2 + 11z + 1$
5	$z^4 + 26z^3 + 66z^2 + 26z + 1$

As $Q_k(z) = z^k Q_k(1/z)$ and $Q_k(0) \neq 0$, $Q_k(z_0) = 0 \iff Q_k(1/z_0) = 0$. Therefore, $Q_k(z)$ has root(s) outside the closed unit disk for all $k \geq 2$.

lf

$$P_{a}(s) = \frac{\omega_{n}^2}{s^2 + \omega_{n}^2} = \frac{j\omega_{n}/2}{s + j\omega_{n}} - \frac{j\omega_{n}/2}{s - j\omega_{n}}$$

then

$$\bar{P}_h(z) = \frac{1}{2} \left(\frac{1 - \mathrm{e}^{-\mathrm{j}\omega_\mathrm{n}h}}{z - \mathrm{e}^{-\mathrm{j}\omega_\mathrm{n}h}} + \frac{1 - \mathrm{e}^{\mathrm{j}\omega_\mathrm{n}h}}{z - \mathrm{e}^{\mathrm{j}\omega_\mathrm{n}h}} \right) = \frac{(1 - \cos(\omega_\mathrm{n}h))(z+1)}{z^2 - 2\cos(\omega_\mathrm{n}h)z + 1}.$$

lf

- $-\cos(\omega_{\mathsf{n}} h)
 eq \pm 1$, then $ar{P}_h(z)$ has two poles at $\mathrm{e}^{\pm \mathrm{j}\omega_{\mathsf{n}} h}$ and a zero at -1,
- $\cos(\omega_{\mathsf{n}} h) = 1$, then $ar{P}_h(z) = 0$,
- $-\cos(\omega_{n}h) = -1$, then $\bar{P}_{h}(z) = 2/(z+1)$.

Thus, even the order of $P_c(s)$ is not always preserved under discretization.

When order drops?

Consider

$$ar{P}_h(z) = \sum_{i=1}^n rac{ar{b}_i}{z - ar{a}_i}$$
 where $ar{a}_i := \mathrm{e}^{a_i h}$ and $ar{b}_i := rac{\mathrm{e}^{a_i h} - 1}{a_i} b_i$.

Two pathological cases, where the order of $\bar{P}_h(z)$ is smaller than n:

1. $\bar{a}_i = \bar{a}_j$, although $a_i \neq a_j$, which is equivalent to

$$\mathrm{e}^{a_i h} = \mathrm{e}^{a_j h} \iff a_i h = a_j h + \mathrm{j} 2\pi k \text{ for some } k \in \mathbb{Z} \setminus \{0\}$$

or
$$a_i - a_j = j2\omega_N k$$
.

or $a_i=\mathrm{j}2\omega_nk$. But if the latter condition holds, then $\exists j
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or
$$a_i - a_j = j2\omega_N k$$
.

2. $\bar{b}_i = 0$, although $b_i \neq 0$, which is equivalent to

$$(e^{a_ih}=1) \wedge (a_i \neq 0) \iff a_ih=j2\pi k \text{ for some } k \in \mathbb{Z} \setminus \{0\}$$

or $a_i = j2\omega_N k$. But if the latter condition holds, then $\exists j \neq i$ such that $a_i = -j2\omega_N k$. Hence, $a_i - a_i = j2\omega_N(2k)$ and this case is covered by 1.

Im s

Pathological sampling

We say that sampling is pathological with respect to P_a if there are at least 2 poles of $P_a(s)$, say p_1 and p_2 , such that

$$p_1 - p_2 = j\frac{2\pi}{h}k = j2\omega_N k \qquad \Longleftrightarrow \qquad \frac{\sum_{k=0}^{\infty} x^k}{\sum_{k=0}^{\infty} x^k} \times \frac{x^k}{\sum_{k=0}^{\infty} x^k}$$

for some $k \in \mathbb{Z} \setminus \{0\}$.

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for some $k \in \mathbb{Z} \setminus \{0\}$. If sampling is pathological, then

some parts of dynamics of P are not visible by the discrete controller.

But these parts don't disappear, they are just in the blind spot of \bar{C} , which cannot counteract anything caused by them (e.g. instability or oscillations).

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$$p_1 - p_2 = j\frac{2\pi}{h}k = j2\omega_N k \iff \frac{\sum_{k=1}^{\infty} k^{p_1}}{\sum_{k=1}^{\infty} k^{p_2}} \times \frac{\mathbb{R}es}{k}$$

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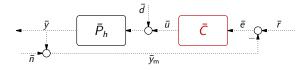
— some parts of dynamics of P are not visible by the *discrete* controller. But these parts don't disappear, they are just in the blind spot of \bar{C} , which cannot counteract anything caused by them (e.g. instability or oscillations).

As the minimum distance between poles for h being pathological is $2\omega_{\rm N}$,

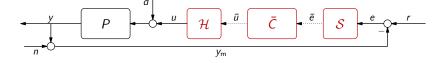
"sufficiently fast" sampling is never pathological.

Fundamental stability result

If sampling is pathological with respect to no unstable poles of $P_{\rm a}(s)$, then \bar{C} stabilizes



iff \bar{C} stabilizes



Outline

Analog redesign

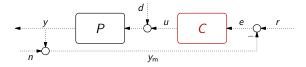
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Discrete unity feedback

We may now drop all signs of discretization and consider a discrete system,

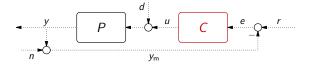


for a given

$$P(z) = \frac{b_m z^m + b_{m-1} z^{m-1} + \dots + b_1 z + b_0}{z^n + a_{n-1} z^{n-1} + \dots + a_1 z + a_0} = \frac{N_P(z)}{D_P(z)}$$

with $b_m \neq 0$ and $m \leq n$ (typically, m = n - 1).

Internal stability



The closed-loop system is said to be

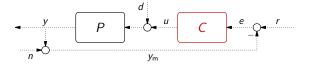
internally stable if all Gang of Four transfer functions

$$\begin{bmatrix} S(z) & T_{d}(z) \\ T_{c}(z) & T(z) \end{bmatrix} := \frac{1}{1 + P(z)C(z)} \begin{bmatrix} 1 \\ C(z) \end{bmatrix} \begin{bmatrix} 1 & P(z) \end{bmatrix}$$

are stable,

i.e. the corresponding transfer function is proper and has no poles outside the open unit disk \mathbb{D} .

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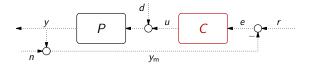
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Internal stability is the formalism helping to avoid unstable cancellations.

Characteristic polynomial



If $C(z) = N_C(z)/D_C(z)$ is proper, then the closed-loop system is internally stable iff its characteristic polynomial

$$\chi_{\rm cl}(z) = N_P(z)N_C(z) + D_P(z)D_C(z)$$

has all roots in \mathbb{D} (such polynomials are known as Schur).

Root locus

The technique is exactly as in the continuous-time case. Start with writing

$$\chi_{\rm cl}(z)=0 \quad \Longleftrightarrow \quad -\frac{1}{k}=G_k(z),$$

where k is a parameter to change, in $(0, \infty)$, and $G_k(z)$ is a proper transfer function. This representation is termed the root-locus form. All rules, which we know from the continuous-time analysis, apply then literally.

— stability / performance areas become different.

For example, no asymptote remains in the stability area (D), which implies that we can afford

- no high-gain feedback in discrete setting if P(z) is strictly proper,

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What changes is the meaning of the results, because

stability / performance areas become different.

For example, no asymptote remains in the stability area (\mathbb{D}) , which implies that we can afford

- no high-gain feedback in discrete setting if P(z) is strictly proper, which is normally the case.

Consider again

$$P(z) = \frac{(h + e^{-h} - 1)z + 1 - (1 + h)e^{-h}}{(z - 1)(z - e^{-h})},$$

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– start:
$$z=1$$
 and $z=\mathrm{e}^{-h}$ (poles of $G_k(z)=P(z)$);

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- end: $z=-\frac{1-(1+h)e^{-h}}{h+e^{-h}-1}\in(-1,0)$ and $z\to-\infty+j0$, as the pole excess is 1 (one asymptote, with the angle 180°);

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$$P(z) = \frac{(h + e^{-h} - 1)z + 1 - (1 + h)e^{-h}}{(z - 1)(z - e^{-h})},$$

which is the discretization of P(s) = 1/[s(s+1)], and the "P" C(z) = k.

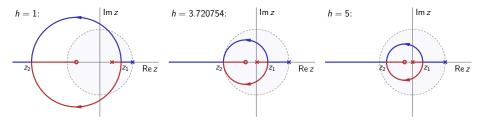
- start: z = 1 and $z = e^{-h}$ (poles of $G_k(z) = P(z)$);
- end: $z=-\frac{1-(1+h)\mathrm{e}^{-h}}{h_{+,o}-h_{-1}}\in(-1,0)$ and $z\to-\infty+\mathrm{j}0$, as the pole excess is 1 (one asymptote, with the angle 180°);
- real axis: between the poles and to the left of the zero
- breakaway / break-in: by dP(z)/dz = 0 for real z,

$$z_{1,2} = \mathrm{e}^{-h} + rac{(1-\mathrm{e}^{-h})\sqrt{1-\mathrm{e}^{-h}}}{\sqrt{1-\mathrm{e}^{-h}} \pm \sqrt{h}} = 1 \mp rac{(1-\mathrm{e}^{-h})\sqrt{h}}{\sqrt{1-\mathrm{e}^{-h}} \pm \sqrt{h}}$$

with $e^{-h} < z_1 < 1$ (breakaway) and z_2 to the left of the zero (break-in) and $z_2 < -1$ if 0 < h < 3.720754 and $-1 < z_1 < 0$ if h > 3.720754.

Root locus: example (contd)

For various sampling periods,

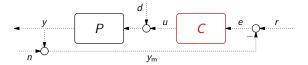


In all cases the system is stable only if k is sufficiently small. In fact, for

$$0 < k < \frac{1 - e^{-h}}{1 - (h+1)e^{-h}}$$

which can be derived by the Jury stability criterion (discrete counterpart of the Routh criterion).

Nyquist criterion



The same logic, as in the continuous-time case. The return difference

$$1 + L(z) = 1 + P(z)C(z) = \frac{\chi_{cl}(z)}{\chi_{ol}(z)}$$

still has open-loop poles as its poles and closed-loop poles as its zeros. The line of reasonings is then

- 1. define simple closed contour Γ_{τ} containing all $\mathbb{C} \setminus \bar{\mathbb{D}}_1$;
- 2. determine the mapping Γ_I of Γ_Z by the loop gain L(Z);
- 3. count the number ν of *clockwise* encirclings of (-1,0) by Γ_L .

By the argument principle, $\nu=\#_{\text{clsd-loop unstable poles}}-\#_{\text{opn-loop unstable poles}}$.

Nyquist contour

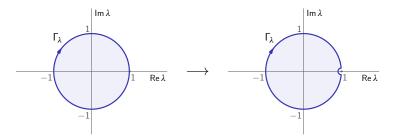
The contour encircling the unstable region $\mathbb{C}\setminus\bar{\mathbb{D}}_1$ is cumbersome. A simple workaround is to redefine $z\to 1/\lambda$. The unstable region in terms of λ is \mathbb{D}_1 and the contour around it is the unit circle, $\Gamma_\lambda=\mathbb{T}$. Some observations:

- the (clockwise) Γ_λ is mapped by $L(\lambda)$ as the frequency response $L(e^{j\theta})$ under increasing θ (the frequency for λ is $-\theta$);

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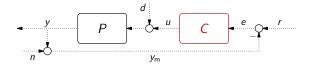
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- the (clockwise) Γ_λ is mapped by $L(\lambda)$ as the frequency response $L(e^{j\theta})$ under increasing θ (the frequency for λ is $-\theta$);
- if $L(\lambda)$, equivalently L(z), has poles at \mathbb{T} , the contour is altered as



with the same completion rules as in the continuous-time case.

Steady-state performance



Nothing changes vis-à-vis the continuous-time case, except replacing s=0 with z=1. For example, if $d[t]=\mathbb{1}[t]$, then by the Final Value Theorem

$$y_{ss} := \lim_{t \to \infty} y[t] = \lim_{z \to 1} (z-1)T_{d}(z)D(z) = \lim_{z \to 1} (z-1)T_{d}(z)\frac{z}{z-1} = T_{d}(1),$$

which is the static gain of (stable) T_d . Moreover,

$$y_{ss} = 0 \quad \iff \quad (P(1) = 0) \lor (|C(1)| = \infty),$$

where the latter condition requires an integral action in C.

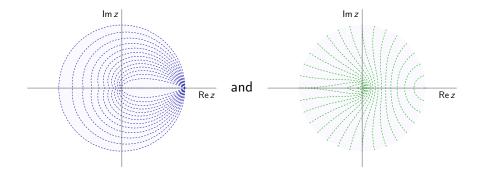
Transient performance and poles

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Because $\mathbb{1} = \mathcal{H}_{ZOH} \bar{\mathbb{1}}$, we have $\mathcal{S}_{idl} G \mathbb{1} = \bar{G}_h \bar{\mathbb{1}}$, i.e. the - step response of the discrete \bar{G}_h is the sampled version of that of G. If $G(s) = \omega_n^2/(s^2 + 2\zeta\omega_n s + \omega_n^2)$ for $\zeta \in [0, 1]$, then $\bar{G}_h(z)$ has its poles at $z={
m e}^{-\xi\omega_{\rm n}h}{
m e}^{\pm{
m j}\sqrt{1-\zeta^2}\omega_{\rm n}h}.$ Constant ζ and $\omega_{\rm n}h$ contours are



Deadbeat control

Given *n*-order P(z) and n_c -order C(z). If the attained

$$\chi_{\rm cl}(z)=z^{n+n_{\rm c}}$$

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We know it as the FIR (finite impulse response) property, impossible in the finite-dimensional continuous-time LTI case. For example, consider

$$S(z) = \frac{1}{1 + P(z)C(z)} = \frac{b_{n+n_c}z^{n+n_c} + b_{n+n_c-1}z^{n+n_c-1} + \dots + b_1z + b_0}{\chi_{cl}(z)}$$
$$= b_{n+n_c} + b_{n+n_c-1}z^{-1} + \dots + b_1z^{1-n-n_c} + b_0z^{-n-n_c}$$

Its impulse response

$$s[t] = b_{n+n_c}\delta[t] + \cdots + b_1\delta[t-n-n_c+1] + b_0\delta[t-n-n_c]$$

indeed ends after at most $n + n_c$ steps.

Deadbeat control: example

Consider

$$P(z) = \frac{h^2}{2} \frac{z+1}{(z-1)^2}$$

which is the discretized $1/s^2$. With $\chi_{cl}(z) = z^3$ we have (see Lecture 5)

$$\begin{bmatrix} 1 & 0 & 0 & 0 \\ -2 & 1 & h^2/2 & 0 \\ 1 & -2 & h^2/2 & h^2/2 \\ 0 & 1 & 0 & h^2/2 \end{bmatrix} \begin{bmatrix} \alpha_1 \\ \alpha_0 \\ \beta_1 \\ \beta_0 \end{bmatrix} = \begin{bmatrix} 1 \\ 0 \\ 0 \\ 0 \end{bmatrix} \implies \begin{bmatrix} \alpha_1 \\ \alpha_0 \\ \beta_1 \\ \beta_0 \end{bmatrix} = \begin{bmatrix} 1 \\ 3/4 \\ 5/(2h^2) \\ -3/(2h^2) \end{bmatrix}$$

so that

$$C(z) = \frac{2}{h^2} \frac{5z - 3}{4z + 3}$$

In this case

$$S(z) = \frac{(z-1)^2(4z+3)}{4z^3} \implies e[t] = \delta[t] - \frac{1}{4}\delta[t-1] - \frac{3}{4}\delta[t-2]$$

with r = 1 (for which $R(z) = \frac{z}{z-1}$ and $S(z)R(z) = 1 - \frac{1}{4}z^{-1} - \frac{3}{4}z^{-2}$).