Control System Toolbox

For Use with MATLAB®

- Computation
- Visualization
- Programming



Version 6



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Control System Toolbox Reference

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Function Reference

Classical Design	Pole Placement LQR/LQG Design State-Space Models Fime Delays Model Dimensions and Characteristics	Matrix Equation Solvers
Classical Design	System Gain and Dynamics Fime Domain Analysis Frequency Domain Analysis Classical Design Pole Placement LQR/LQG Design State-Space Models	·
Classical Design	System Gain and Dynamics Fime Domain Analysis Frequency Domain Analysis Classical Design	State-Space Models
	System Gain and Dynamics	Classical Design

Function Reference

 $Functions - Categorical \ List (p. \ 1-2) \qquad Lists \ the \ Control \ System \ Toolbox \ functions \ according \ to$

their purpose.

Functions — Alphabetical List (p. 1-9) Lists the Control System Toolbox functions

alphabetically.

Functions — Categorical List

General

ctrlpref Set Control System Toolbox preferences
ltimodels Detailed help on the various types of LTI

models

1tiprops Detailed help on available LTI model properties

Creating Linear Models

filt Specify a digital filter

frd Create a frequency-response data models

lti/set Set/modify properties of LTI models

ss, dss Create state-space models (continuous/discrete)

tf Create transfer function models
zpk Create zero/pole/gain models

Data Extraction

dssdata Descriptor version of ssdata

frdata Extract frequency-response data

lti/get Access values of LTI model properties

ssdata Extract state-space matrices

tfdata Extract numerators and denominators

zpkdata Extract zero/pole/gain data

Conversions

c2d Convert from continuous- to discrete-time models

chunits Convert the units property for FRD models

d2c Convert from discrete- to continuous-time models

d2d Test true for continuous-time models

frd Convert to a frequency-response data model

SS Convert to a state-space model

tf Convert to a transfer function model
zpk Convert to a zero/pole/gain model

System Interconnections

append Group LTI systems by appending inputs and outputs

connect Derive state-space models from block diagram

descriptions

feedback Feedback connections of two systems

1ft Generalized feedback interconnection (Redheffer star

product)

parallel Generalized parallel connection (see also overloaded

+)

series Generalized series connection (see also overloaded *)

System Gain and Dynamics

bandwidth System bandwidth

dcgain D.C. (low-frequency) gain

bandwidth System bandwidth

damp Natural frequency and damping of system poles

dsort Norms of LTI systems

esort Sort continuous poles by real part

iopzmap Input/output pole/zero map

lti/norm Norms of LTI systems

modsep Region-based modal decomposition

pole, eig System poles
pzmap Pole/zero map

stabsep Stable/unstable decomposition

Time Domain Analysis

covar Covariance of response to white noise

gensig Generate input signal for 1sim

impulse Impulse response

initial Response of state-space system with given initial

state

1sim Response to arbitrary inputs

ltiview Response analysis GUI (LTI Viewer)

step Step response

Frequency Domain Analysis

allmargin All crossover frequencies and related gain/phase

margins

bode Bode diagrams of the frequency response

bodemag Bode magnitude diagram only

evalfr Evaluate frequency response at given frequency

frequency response over a frequency grid

frd/interp Interpolate frequency-esponse data

ltiview Response analysis GUI (LTI Viewer)

margin Gain and phase margins

nichols Nichols plot

nyquist Nyquist plot

sigma Plot the pole/zero map of an LTI model

Classical Design

rlocus Evans root locus

sisotool SISO design GUI (root locus and loop-shaping

techniques)

Pole Placement

acker SISO pole placement

estim Form estimator given estimator gain

place MIMO pole placement

reg Form regulator given state-feedback and estimator

gain

LQR/LQG Design

augstate Augment output by appending states

lqg Single-step LQG design

lgr, dlqr Linear-quadratic (LQ) state-feedback regulator lqrd Discrete LQ regulator for continuous plants

1qrreg Form LQG regulator given LQ gain and Kalman

estimator

1qgy LQ regulator with output weighting

kalman Kalman estimator

kalmand Discrete Kalman estimator for continuous plants

State-Space Models

balreal Grammian-based input/output balancing

canon State-space canonical forms

ctrb Controllability matrix

gram Controllability and observability grammians
minreal Minimal realization and pole/zero cancellation

modred Model state reduction

margin Calculate gain and phase margins

ngrid Superimpose grid lines on a Nichols plot

nichols Calculate Nichols plot
nyquist Calculate Nyquist plot
obsv Observability matrix

sminreal Structurally minimal realization ss2ss State coordinate transformation

ssbal Diagonal balancing of state-space realizations

Time Delays

delay2z Replace delays by poles at z=0 or FRD phase shift

hasdelay True for models with time delays
pade Pade approximation of time delays

totaldelay Total delay between each input/output pair

Model Dimensions and Characteristics

class Model type ('tf', 'zpk', 'ss', or 'frd')

isct True for continuous-time models

isdt True for discrete-time models

isproper True for proper models

issiso True for single-input/single-output models

lti/ndims Number of dimensions

 ${\tt lti/isempty} \qquad \qquad {\tt True} \ {\tt for} \ {\tt empty} \ {\tt LTI} \ {\tt models}$

reshape Reshape array of linear models

size Model sizes and order

Overloaded and Arithmetic Operators

+ and - Add and subtract systems (parallel connection)

* Multiply systems (series connection)

\ Left divide — sys1\sys2 means inv(sys1)*sys2

Right divide — sys1/sys2 means sys1*inv(sys2)

[^] Powers of a given system

' Pertransposition

. ' Transposition of input/output map

[..] Concatenate models along inputs or outputs

stack Stack models/arrays along some array dimension

lti/inv Inverse of an LTI system

conj Complex conjugation of model coefficients

Matrix Equation Solvers

bdschur Block diagonalization of a square matrix

care, dare Solve algebraic Riccati equations

gcare, gdare Solve generalized algebraic Riccati equations

lyap, dlyap Solve Lyapunov equations

lyapchol, Square-root Lyapunov equations

dlyapschol

Command-Line Plot Customization

bodeplot Bode magnitude and phase plus plot handle

getoptions Get the plot options handle

hsvplot Hankel singular value plus plot handle

impulseplot Impulse response plus plot handle

initialplot Initial condition plus plot handle

iopzplot Pole/zero maps for input/output pairs plus plot

handle

1simplot Time response to arbitrary inputs plus plot

handle

nicholsplot Nichols plot plus plot handle

nyquistplot Nyquist plus plot handle

pzplot Pole/zero plus plot handle

rlocusplot Root locus plus plot handle

setoptions Set plot options

sigmaplot Singular values of the frequency response plus

plot handle

stepplot Step response plus plot handle

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Purpose P

Pole placement design for single-input systems

Syntax

k = acker(A,b,p)

Description

Given the single-input system

$$\dot{x} = Ax + bu$$

and a vector p of desired closed-loop pole locations, acker (A,b,p) uses Ackermann's formula [1] to calculate a gain vector k such that the state feedback u=-kx places the closed-loop poles at the locations p. In other words, the eigenvalues of A-bk match the entries of p (up to ordering). Here A is the state transmitter matrix and b is the input to state transmission vector.

You can also use acker for estimator gain selection by transposing the matrix A and substituting c' for b when y = cx is a single output.

Limitations

acker is limited to single-input systems and the pair (A, b) must be controllable.

Note that this method is not numerically reliable and starts to break down rapidly for problems of order greater than 5 or for weakly controllable systems. See place for a more general and reliable alternative.

References

[1] Kailath, T., Linear Systems, Prentice-Hall, 1980, p. 201.

See Also

lqrOptimal LQ regulatorplacePole placement designrlocusRoot locus design

allmargin

Purpose

Compute all crossover frequencies and corresponding stability margins

Syntax

S = allmargin(sys)

Description

allmargin computes the gain, phase, and delay margins and the corresponding crossover frequencies of the SISO open-loop model sys. allmargin is applicable to any SISO model, including models with delays.

The output S is a structure with the following fields:

- GMFrequency All -180 degree crossover frequencies (in rad/sec)
- GainMargin Corresponding gain margins, defined as 1/G where G is the gain at crossover
- PMFrequency All 0 dB crossover frequencies in rad/sec
- PhaseMargin Corresponding phase margins in degrees
- DMFrequency and DelayMargin Critical frequencies and the corresponding delay margins. Delay margins are given in seconds for continuous-time systems and multiples of the sample time for discrete-time systems.
- Stable 1 if the nominal closed-loop system is stable, 0 otherwise.

In general, stability cannot be assessed for FRD system. In any case when stability cannot be assessed, S is set to NaN.

See Also

 $\begin{array}{ll} \hbox{\tt ltimodels} & & \hbox{\tt Help on LTI models} \\ \hbox{\tt ltiview} & & \hbox{\tt LTI system viewer} \end{array}$

margin Gain and phase margins for SISO open-loop systems

Purpose

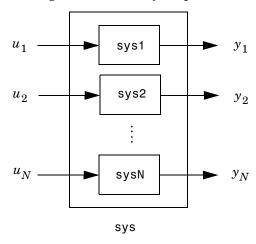
Group LTI models by appending their inputs and outputs

Syntax

$$sys = append(sys1, sys2, ..., sysN)$$

Description

append appends the inputs and outputs of the LTI models sys1,...,sysN to form the augmented model sys depicted below.



For systems with transfer functions $H_1(s)$,..., $H_N(s)$, the resulting system sys has the block-diagonal transfer function

$$\begin{bmatrix} H_1(s) & 0 & \dots & 0 \\ 0 & H_2(s) & \dots & \vdots \\ \vdots & \ddots & \ddots & 0 \\ 0 & \dots & 0 & H_N(s) \end{bmatrix}$$

For state-space models sys1 and sys2 with data (A_1,B_1,C_1,D_1) and (A_2,B_2,C_2,D_2) , append(sys1,sys2) produces the following state-space model.

append

$$\begin{bmatrix} \dot{x}_1 \\ \dot{x}_2 \end{bmatrix} = \begin{bmatrix} A_1 & 0 \\ 0 & A_2 \end{bmatrix} \begin{bmatrix} x_1 \\ x_2 \end{bmatrix} + \begin{bmatrix} B_1 & 0 \\ 0 & B_2 \end{bmatrix} \begin{bmatrix} u_1 \\ u_2 \end{bmatrix}$$
$$\begin{bmatrix} y_1 \\ y_2 \end{bmatrix} = \begin{bmatrix} C_1 & 0 \\ 0 & C_2 \end{bmatrix} \begin{bmatrix} x_1 \\ x_2 \end{bmatrix} + \begin{bmatrix} D_1 & 0 \\ 0 & D_2 \end{bmatrix} \begin{bmatrix} u_1 \\ u_2 \end{bmatrix}$$

Arguments

The input arguments sys1,..., sysN can be LTI models of any type. Regular matrices are also accepted as a representation of static gains, but there should be at least one LTI object in the input list. The LTI models should be either all continuous, or all discrete with the same sample time. When appending models of different types, the resulting type is determined by the precedence rules (see Precedence Rules for details).

There is no limitation on the number of inputs.

Example

The commands

produce the state-space model

sys a = x2 х1 0 0 х1 x2 0 1.00000 b =u1 u2 u3 1.00000 0 х1 x2 0 0 2.00000 c = х2 х1 1.00000 0 y1

append

	y2 y3	0	0 3.00000	
d =				
		u1	u2	u3
	y1	0	0	0
	y2	0	10.00000	0
	v3	0	0	4.00000

Continuous-time system.

Se	е	ΑI	SO

Modeling of block diagram interconnections
Feedback connection
Parallel connection
Series connection

augstate

Purpose

Append the state vector to the output vector

Syntax

asys = augstate(sys)

Description

Given a state-space model sys with equations

$$\dot{x} = Ax + Bu$$

$$y = Cx + Du$$

(or their discrete-time counterpart), augstate appends the states \boldsymbol{x} to the outputs \boldsymbol{y} to form the model

$$\dot{x} = Ax + Bu$$

$$\begin{bmatrix} y \\ x \end{bmatrix} = \begin{bmatrix} C \\ I \end{bmatrix} x + \begin{bmatrix} D \\ 0 \end{bmatrix} u$$

This command prepares the plant so that you can use the feedback command to close the loop on a full-state feedback u = -Kx.

Limitation

Because augstate is only meaningful for state-space models, it cannot be used with TF, ZPK or FRD models.

See Also

feedbackFeedback connectionparallelParallel connectionseriesSeries connection

Purpose

Gramian-based input/output balancing of state-space realizations

Syntax

Description

[sysb,g] = balreal(sys) computes a balanced realization sysb for the stable portion of the LTI model sys. balreal handles both continuous and discrete systems. If sys is not a state-space model, it is first and automatically converted to state space using ss.

For stable systems, sysb is an equivalent realization for which the controllability and observability Gramians are equal and diagonal, their diagonal entries forming the vector G of Hankel singular values. Small entries in G indicate states that can be removed to simplify the model (use modred to reduce the model order).

If sys has unstable poles, its stable part is isolated, balanced, and added back to its unstable part to form sysb. The entries of g corresponding to unstable modes are set to Inf. You can specify additional options for the stable/unstable decomposition:

See stabsep for more details on these options. The default values are ATOL=0, RTOL=1e-8, and ALPHA=1e-8.

Use balreal(sys,condmax) to control the condition number of the stable/ unstable decomposition. Increasing condmax helps separate close by stable and unstable modes at the expense of accuracy. By default condmax=1e8.

[sysb,g,T,Ti] = balreal(sys) also returns the vector g containing the diagonal of the balanced gramian, the state similarity transformation $x_h = Tx$

used to convert sys to sysb, and the inverse transformation $T_i = T^{-1}$.

If the system is normalized properly, the diagonal g of the joint gramian can be used to reduce the model order. Because g reflects the combined controllability and observability of individual states of the balanced model, you can delete those states with a small g(i) while retaining the most important input-output characteristics of the original system. Use modred to perform the state elimination.

There are also overloaded methods available. Type

```
help ss/balreal
help lti/balreal
help idmodel/balreal
```

for more information.

Example 1

Consider the zero-pole-gain model

A state-space realization with balanced gramians is obtained by

```
[sysb,g] = balreal(sys)
```

The diagonal entries of the joint gramian are

```
g'
ans =
1.0062e-01 6.8039e-05 1.0055e-05
```

which indicates that the last two states of sysb are weakly coupled to the input and output. You can then delete these states by

```
sysr = modred(sysb,[2 3],'del')
```

to obtain the following first-order approximation of the original system.

```
zpk(sysr)
```

Zero/pole/gain:

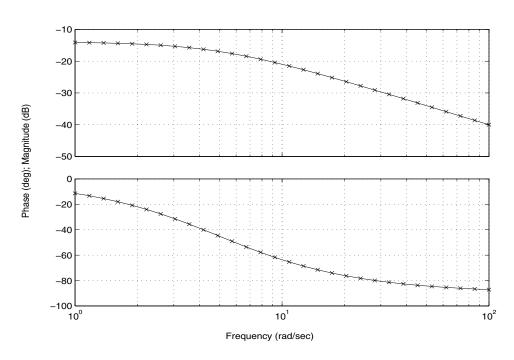
1.0001

(s+4.97)

Compare the Bode responses of the original and reduced-order models.

bode(sys,'-',sysr,'x')

Bode Diagrams



Example 2

Create this unstable system:

sys1=tf(1,[1 0 -1])

Transfer function:

, -----

balreal

$$s^2 - 1$$

Apply balreal to create a balanced gramian realization.

[sysb,g]=balreal(sys1)

Continuous-time model.

The unstable pole shows up as Inf in vector g.

Algorithm

Consider the model

$$\dot{x} = Ax + Bu$$
$$y = Cx + Du$$

with controllability and observability gramians W_c and W_o . The state coordinate transformation $\bar{x}=Tx$ produces the equivalent model

$$\dot{\bar{x}} = TAT^{-1}\bar{x} + TBu$$

$$y = CT^{-1}\overline{x} + Du$$

and transforms the gramians to

$$\overline{W}_c = TW_cT^T, \qquad \overline{W}_o = T^{-T}W_oT^{-1}$$

The function balreal computes a particular similarity transformation T such that

$$\overline{W}_c = \overline{W}_o = diag(g)$$

See [1,2] for details on the algorithm.

References

[1] Laub, A.J., M.T. Heath, C.C. Paige, and R.C. Ward, "Computation of System Balancing Transformations and Other Applications of Simultaneous Diagonalization Algorithms," *IEEE Trans. Automatic Control*, AC-32 (1987), pp. 115-122.

[2] Moore, B., "Principal Component Analysis in Linear Systems: Controllability, Observability, and Model Reduction," *IEEE Transactions on Automatic Control*, AC-26 (1981), pp. 17-31.

[3] Laub, A.J., "Computation of Balancing Transformations," *Proc. ACC*, San Francisco, Vol.1, paper FA8-E, 1980.

See Also

gram	Controllability and observability gramians
modred	Model order reduction
SS	Convert LTI model to state space
ssbal	Balancing of state-space model using diagonal

similarity

balred

Purpose

Model order reduction

Syntax

```
rsys = balred(sys,ORDERS)
rsys = balred(sys,ORDERS,...,'Elimination',METHOD)
rsys = balred(sys,ORDERS,...,'Balancing',BALDATA)
```

Description

rsys = balred(sys,ORDERS) computes a reduced-order approximation rsys of the LTI model sys. The desired order (number of states) for rsys is specified by ORDERS. You can try multiple orders at once by setting ORDERS to a vector of integers, in which case rsys is a vector of reduced-order models. Use hsvd to plot the Hankel singular values and pick an adequate approximation order. States with relatively small Hankel singular values can be safely discarded.

When sys has unstable poles, it is first decomposed into its stable and unstable parts using stabsep, and only the stable part is approximated. Use

to specify additional options for the stable/unstable decomposition. See stabsep for details. The default values are ATOL=0, RTOL=1e-8, and ALPHA=1e-8.

rsys = balred(sys,ORDERS,...,'Elimination',METHOD) specifies the state elimination method. Available choices for METHOD include:

- 'MatchDC': Enforce matching DC gains (default)
- 'Truncate': Simply discard the states associated with small Hankel singular values. The 'Truncate' method tends to produce a better approximation in the frequency domain, but the DC gains are not guaranteed to match.

rsys = balred(sys,ORDERS,...,'Balancing',BALDATA) makes use of the balancing data BALDATA produced by hsvd. Because hsvd does most of the work needed to compute rsys, this syntax is more efficient when using hsvd and balred jointly.

balred uses implicit balancing techniques to compute the reduced- order approximation rsys.

There is more than one balred method available. Type

help lti/balred

for more information.

References [1] Varga, A., "Balancing-Free Square-Root Algorithm for Computing Singular

Perturbation Approximations," Proc. of 30th IEEE CDC, Brighton, UK (1991),

pp. 1062-1065.

See Also hvsd Computes the Hankel singular values of an LTI model

lti/order LTI model order

minreal Minimal realization and pole-zero cancellation sminreal Compute a structurally minimal realization

bandwidth

Purpose Compute the frequency response bandwidth

Syntax fb = bandwidth(sys)

fb = bandwidth(sys,dbdrop)

Description fb = bandwidth(sys) computes the bandwidth fb of the SISO model sys,

defined as the first frequency where the gain drops below 70.79 percent (-3 dB) of its DC value. The frequency ${\tt fb}$ is expressed in radians per second. You can

create sys using tf, ss, or zpk. See ltimodels for details.

fb = bandwidth(sys,dbdrop) further specifies the critical gain drop in dB.

The default value is -3 dB, or a 70.79 percent drop.

If sys is an S1-by...-by-Sp array of LTI models, bandwidth returns an array of $\,$

the same size such that

fb(j1,...,jp) = bandwidth(sys(:,:,j1,...,jp))

See Also dcgain Compute the steady-state gain of LTI models

issiso Returns 1 if the system is SISO

ltimodels General information about LTI models

Purpose

Compute the Bode frequency response of LTI models

Syntax

```
bode(sys)
bode(sys,w)

bode(sys1,sys2,...,sysN)
bode(sys1,sys2,...,sysN,w)
bode(sys1,'PlotStyle1',...,sysN,'PlotStyleN')

[mag,phase,w] = bode(sys)
```

Description

bode computes the magnitude and phase of the frequency response of LTI models. When invoked without left-side arguments, bode produces a Bode plot on the screen. The magnitude is plotted in decibels (dB), and the phase in degrees. The decibel calculation for mag is computed as $20\log_{10}(|H(j\omega)|)$, where $|H(j\omega)|$ is the system's frequency response. Bode plots are used to analyze system properties such as the gain margin, phase margin, DC gain, bandwidth, disturbance rejection, and stability.

bode (sys) plots the Bode response of an arbitrary LTI model sys. This model can be continuous or discrete, and SISO or MIMO. In the MIMO case, bode produces an array of Bode plots, each plot showing the Bode response of one particular I/O channel. The frequency range is determined automatically based on the system poles and zeros.

bode(sys,w) explicitly specifies the frequency range or frequency points to be used for the plot. To focus on a particular frequency interval [wmin,wmax], set w = {wmin,wmax}. To use particular frequency points, set w to the vector of desired frequencies. Use logspace to generate logarithmically spaced frequency vectors. All frequencies should be specified in radians/sec.

bode (sys1,sys2,...,sysN) or bode (sys1,sys2,...,sysN,w) plots the Bode responses of several LTI models on a single figure. All systems must have the same number of inputs and outputs, but may otherwise be a mix of continuous and discrete systems. This syntax is useful to compare the Bode responses of multiple systems.

bode(sys1, 'PlotStyle1',...,sysN, 'PlotStyleN') specifies which color, linestyle, and/or marker should be used to plot each system. For example,

uses red dashed lines for the first system sys1 and green 'x' markers for the second system sys2.

When invoked with left-side arguments

```
[mag,phase,w] = bode(sys)
[mag,phase] = bode(sys,w)
```

return the magnitude and phase (in degrees) of the frequency response at the frequencies w (in rad/sec). The outputs mag and phase are 3-D arrays with the frequency as the last dimension (see "Arguments" below for details). You can convert the magnitude to decibels by

$$magdb = 20*log10(mag)$$

Remark

If sys is an FRD model, bode(sys,w), w can only include frequencies in sys.frequency.

Arguments

The output arguments mag and phase are 3-D arrays with dimensions

```
(number of outputs) × (number of inputs) × (length of w)
```

For SISO systems, mag(1,1,k) and phase(1,1,k) give the magnitude and phase of the response at the frequency $\omega_k = w(k)$.

$$\max(1,1,k) = |h(j\omega_k)|$$

$$\text{phase}(1,1,k) = \angle h(j\omega_k)$$

MIMO systems are treated as arrays of SISO systems and the magnitudes and phases are computed for each SISO entry h_{ij} independently $(h_{ij}$ is the transfer function from input j to output i). The values mag(i,j,k) and phase(i,j,k) then characterize the response of h_{ij} at the frequency w(k).

$$\begin{aligned} \text{mag(i,j,k)} &= \left| h_{ij}(j\omega_k) \right| \\ \text{phase(i,j,k)} &= \angle h_{ij}(j\omega_k) \end{aligned}$$

Example

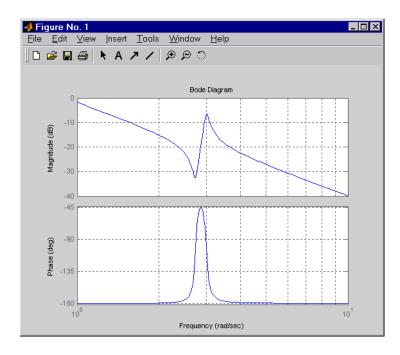
You can plot the Bode response of the continuous SISO system

$$H(s) = \frac{s^2 + 0.1s + 7.5}{s^4 + 0.12s^3 + 9s^2}$$

by typing

$$g = tf([1 \ 0.1 \ 7.5],[1 \ 0.12 \ 9 \ 0 \ 0]);$$

bode(g)

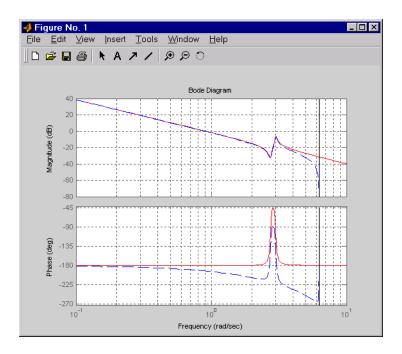


To plot the response on a wider frequency range, for example, from 0.1 to 100 rad/sec, type

$$bode(g, \{0.1, 100\})$$

You can also discretize this system using zero-order hold and the sample time $T_s=0.5\,$ second, and compare the continuous and discretized responses by typing

$$gd = c2d(g,0.5)$$



Algorithm

For continuous-time systems, bode computes the frequency response by evaluating the transfer function H(s) on the imaginary axis $s=j\omega$. Only positive frequencies ω are considered. For state-space models, the frequency response is $D+C(j\omega-A)^{-1}B$, $\omega \geq 0$

When numerically safe, A is diagonalized for maximum speed. Otherwise, A is reduced to upper Hessenberg form and the linear equation $(j\omega - A)X = B$ is solved at each frequency point, taking advantage of the Hessenberg structure. The reduction to Hessenberg form provides a good compromise between efficiency and reliability. See [1] for more details on this technique.

For discrete-time systems, the frequency response is obtained by evaluating the transfer function H(z) on the unit circle. To facilitate interpretation, the upper-half of the unit circle is parametrized as

$$z=e^{j\omega T_s}, \qquad 0\leq \omega \leq \omega_N=\frac{\pi}{T_s}$$

where T_s is the sample time. ω_N is called the *Nyquist frequency*. The equivalent "continuous-time frequency" ω is then used as the x-axis variable. Because

$$H(e^{j\omega T_s})$$

is periodic with period $2\omega_N$, bode plots the response only up to the Nyquist frequency ω_N . If the sample time is unspecified, the default value $T_s=1$ is assumed.

Diagnostics

If the system has a pole on the $j\omega$ axis (or unit circle in the discrete case) and w happens to contain this frequency point, the gain is infinite, $j\omega I-A$ is singular, and bode produces the warning message

Singularity in freq. response due to jw-axis or unit circle pole.

References

[1] Laub, A.J., "Efficient Multivariable Frequency Response Computations," *IEEE Transactions on Automatic Control*, AC-26 (1981), pp. 407-408.

See Also

evalfr Response at single complex frequency
freqresp Frequency response computation
LTI system viewer
nichols Nichols plot
nyquist Nyquist plot

sigma Singular value plot

bodemag

Purpose

Compute the Bode magnitude response of LTI models

Syntax

```
bodemag(sys)
```

bodemag(sys,{wmin,wmax})

bodemag(sys,w)

bodemag(sys1,sys2,...,sysN,w)

bodemag(sys1, 'PlotStyle1',...,sysN, 'PlotStyleN')

Description

bodemag(sys) plots the magnitude of the frequency response of the LTI model SYS (Bode plot without the phase diagram). The frequency range and number of points are chosen automatically.

bodemag(sys, {wmin, wmax}) draws the magnitude plot for frequencies between wmin and wmax (in radians/second).

bodemag(sys,w) uses the user-supplied vector W of frequencies, in radians/second, at which the frequency response is to be evaluated.

bodemag(sys1,sys2,...,sysN,w) shows the frequency response magnitude of several LTI models sys1,sys2,...,sysN on a single plot. The frequency vector wis optional. You can also specify a color, line style, and marker for each model, as in

See Also

bode Compute the Bode frequency response of LTI models

ltiview Open an LTI Viewer ltimodels Help on LTI models

Compute the Bode frequency response and return the plot handle

Syntax

```
h = bodeplot(sys1,sys2,...)
h = bodeplot(AX,...)
h = bodeplot(..., plotoptions)
h = bodeplot(sys,w)
```

Description

h = bodeplot(sys) plot the Bode magnitude and phase of an LTI model sys and returns the plot handle h to the plot. You can use this handle to customize the plot with the getoptions and setoptions commands. Type

```
help bodeoptions
```

h = bodeplot(svs)

for a list of available plot options.

bodeplot(sys) draws the Bode plot of the LTI model sys (created with either tf, zpk, ss, or frd). The frequency range and number of points are chosen automatically.

bodeplot(sys1,sys2,...) graphs the Bode response of multiple LTI models sys1,sys2,... on a single plot. You can specify a color, line style, and marker for each model, as in

```
bodeplot(sys1, 'r', sys2, 'y--', sys3, 'gx')
```

bodeplot(AX,...) plots into the axes with handle AX.

bodeplot(..., plotoptions) plots the Bode response with the options specified in plotoptions.

bodeplot(sys,w) draws the Bode plot for frequencies specified by w. When w = {wmin,wmax}, the Bode plot is drawn for frequencies between wmin and wmax (in rad/s). When w is a user-supplied vector w of frequencies, in rad/s, the Bode response is drawn for the specified frequencies.

See logspace to generate logarithmically spaced frequency vectors.

Example

Use the plot handle to change options in a Bode plot.

```
sys = rss(5);
h = bodeplot(sys);
```

bodeplot

% Change units to Hz and make phase plot invisible setoptions(h,'FreqUnits','Hz','PhaseVisible','off');

See Also

bode Bode plots (does not return the plot handle)

getoptions Get plot options from a plot

setoptions Set plot options

Discretize continuous-time systems

Syntax

sysd = c2d(sys,Ts)

sysd = c2d(sys,Ts,method)
[sysd,G] = c2d(sys,Ts,method)

Description

sysd = c2d(sys,Ts) discretizes the continuous-time LTI model sys using zero-order hold on the inputs and a sample time of Ts seconds.

sysd = c2d(sys,Ts,method) gives access to alternative discretization schemes. The string method selects the discretization method among the following:

'zoh' Zero-order hold. The control inputs are assumed piecewise

constant over the sampling period Ts.

'foh' Triangle approximation (modified first-order hold, see [1], p.

151). The control inputs are assumed piecewise linear over

the sampling period Ts.

'tustin' Bilinear (Tustin) approximation.

'prewarp' Tustin approximation with frequency prewarping. You must

specify the critical frequency Wc (in rad/sec) as a fourth input

as in

sysd = c2d(sysc,ts,'prewarp',Wc)

'matched' Matched pole-zero method. See [1], p. 147.

Refer to "Continuous/Discrete Conversions of LTI Models" for more detail on these discretization methods.

c2d supports MIMO systems (except for the 'matched' method) as well as LTI models with delays with some restrictions for 'matched' and 'tustin' methods.

[sysd,G] = c2d(sys,Ts,method) returns a matrix G that maps the continuous initial conditions x_0 and u_0 to their discrete counterparts x[0] and u[0] according to

$$x[0] = G \cdot \begin{bmatrix} x_0 \\ u_0 \end{bmatrix}$$
$$u[0] = u_0$$

Example

Consider the system

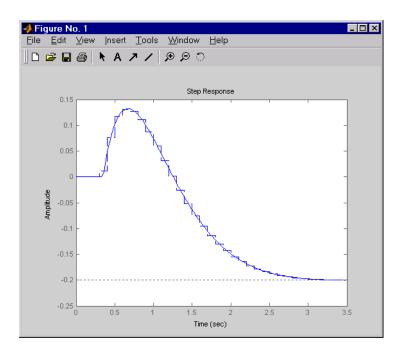
$$H(s) = \frac{s-1}{s^2 + 4s + 5}$$

with input delay T_d = 0.35 second. To discretize this system using the triangle approximation with sample time T_s = 0.1 second, type

Sampling time: 0.1

The next command compares the continuous and discretized step responses.

step(H,'-',Hd,'--')



References

[1] Franklin, G.F., J.D. Powell, and M.L. Workman, *Digital Control of Dynamic Systems*, Second Edition, Addison-Wesley, 1990.

See Also

d2c Discrete to continuous conversion d2d Resampling of discrete systems-.

Compute canonical state-space realizations

Syntax

Description

canon computes a canonical state-space model for the continuous or discrete LTI system sys. Two types of canonical forms are supported.

Modal Form

csys = canon(sys, 'type') returns a realization csys in modal form, that is, where the real eigenvalues appear on the diagonal of the A matrix and the complex conjugate eigenvalues appear in 2-by-2 blocks on the diagonal of A. For a system with eigenvalues $(\lambda_1, \sigma \pm j\omega, \lambda_2)$, the modal A matrix is of the form

$$\begin{bmatrix} \lambda_1 & 0 & 0 & 0 \\ 0 & \sigma & \omega & 0 \\ 0 & -\omega & \sigma & 0 \\ 0 & 0 & 0 & \lambda_2 \end{bmatrix}$$

Companion Form

csys = canon(sys, 'type') produces a companion realization of sys where the characteristic polynomial of the system appears explicitly in the rightmost column of the A matrix. For a system with characteristic polynomial

$$p(s) = s^{n} + a_{1}s^{n-1} + \dots + a_{n-1}s + a_{n}$$

the corresponding companion A matrix is

For state-space models sys,

$$[csys,T] = canon(a,b,c,d,'type')$$

also returns the state coordinate transformation T relating the original state vector x and the canonical state vector x_c .

$$x_c = Tx$$

This syntax returns T=[] when sys is not a state-space model.

Algorithm

Transfer functions or zero-pole-gain models are first converted to state space using ss.

The transformation to modal form uses the matrix P of eigenvectors of the Amatrix. The modal form is then obtained as

$$\dot{x_c} = P^{-1}APx_c + P^{-1}Bu$$
$$y = CPx_c + Du$$

The state transformation T returned is the inverse of P.

The reduction to companion form uses a state similarity transformation based on the controllability matrix [1].

Limitations

The modal transformation requires that the A matrix be diagonalizable. A sufficient condition for diagonalizability is that A has no repeated eigenvalues.

The companion transformation requires that the system be controllable from the first input. The companion form is often poorly conditioned for most state-space computations; avoid using it when possible.

Controllability matrix

References

[1] Kailath, T. Linear Systems, Prentice-Hall, 1980.

See Also

ctrb	Controllability matrix
ctrbf	Controllability canonical form
ss2ss	State similarity transformation

Solve continuous-time algebraic Riccati equations

Syntax

Description

[X,L,G] = care(A,B,Q) computes the unique solution X of the continuous-time algebraic Riccati equation

$$A^T X + XA - XBB^T X + Q = 0$$

The care function also returns the gain matrix, $G = R^{-1}B^T X E$.

[X,L,G] = care(A,B,Q,R,S,E) solves the more general Riccati equation

$$A^{T}XE + E^{T}XA - (E^{T}XB + S)R^{-1}(B^{T}XE + S^{T}) + Q = 0$$

When omitted, R, S, and E are set to the default values R=I, S=0,and E=I. Along with the solution X, care returns the gain matrix $G = R^{-1}(B^TXE + S^T)$ and a vector L of closed-loop eigenvalues, where

[X,L,G,report] = care(A,B,Q, ...) returns a diagnosis report with:

- -1 when the associated Hamiltonian pencil has eigenvalues on or very near the imaginary axis (failure)
- \bullet –2 when there is no finite stabilizing solution X
- The Frobenius norm of the relative residual if X exists and is finite.

This syntax does not issue any error message when X fails to exist.

[X1,X2,D,L] = care(A,B,Q,..., factor) returns two matrices X1, X2 and a diagonal scaling matrix D such that X = D*(X2/X1)*D.

The vector L contains the closed-loop eigenvalues. All outputs are empty when the associated Hamiltonian matrix has eigenvalues on the imaginary axis.

Examples

Example 1

Given

$$A = \begin{bmatrix} -3 & 2 \\ 1 & 1 \end{bmatrix} \qquad B = \begin{bmatrix} 0 \\ 1 \end{bmatrix} \qquad C = \begin{bmatrix} 1 & -1 \end{bmatrix} \qquad R = 3$$

you can solve the Riccati equation

$$A^T X + XA - XBR^{-1}B^T X + C^T C = 0$$

```
by
    a = [-3 2;1 1]
    b = [0 ; 1]
    c = [1 -1]
    r = 3
    [x,1,g] = care(a,b,c'*c,r)
```

This yields the solution

You can verify that this solution is indeed stabilizing by comparing the eigenvalues of a and a-b*g.

```
[eig(a) eig(a-b*g)]
ans =
    -3.4495    -3.5026
    1.4495    -1.4370
```

Finally, note that the variable 1 contains the closed-loop eigenvalues eig(a-b*g).

```
1
1 =
-3.5026
```

-1.4370

Example 2

To solve the H_{∞} -like Riccati equation

$$A^{T}X + XA + X(\gamma^{-2}B_{1}B_{1}^{T} - B_{2}B_{2}^{T})X + C^{T}C = 0$$

rewrite it in the care format as

$$A^{T}X + XA - X \underbrace{\begin{bmatrix} B_1, B_2 \end{bmatrix}}_{B} \underbrace{\begin{bmatrix} -\gamma^{-2}I & 0 \\ 0 & I \end{bmatrix}}^{-1} \begin{bmatrix} B_1^{T} \\ B_2^{T} \end{bmatrix} X + C^{T}C = 0$$

You can now compute the stabilizing solution X by

Algorithm

care implements the algorithms described in [1]. It works with the Hamiltonian matrix when R is well-conditioned and E = I; otherwise it uses the extended Hamiltonian pencil and QZ algorithm.

Limitations

The (A,B) pair must be stabilizable (that is, all unstable modes are controllable). In addition, the associated Hamiltonian matrix or pencil must have no eigenvalue on the imaginary axis. Sufficient conditions for this to hold are (Q,A) detectable when S=0 and R>0, or

$$\begin{bmatrix} Q & S \\ S^T & R \end{bmatrix} > 0$$

References

[1] Arnold, W.F., III and A.J. Laub, "Generalized Eigenproblem Algorithms and Software for Algebraic Riccati Equations," *Proc. IEEE*, 72 (1984), pp. 1746-1754.

See Also

dare lyap Solve discrete-time Riccati equations Solve continuous-time Lyapunov equations

chgunits

Purpose

Convert the frequency units of an FRD model

Syntax

sys = chgunits(sys,units)

Description

sys = chgunits(sys,units) converts the units of the frequency points stored in an FRD model, sys to units, where units is either of the strings 'Hz' or 'rad/s'. This operation changes the assigned frequencies by applying the appropriate (2*pi) scaling factor, and the 'Units' property is updated.

If the 'Units' field already matches units, no conversion is made.

Example

```
w = logspace(1,2,2);
sys = rss(3,1,1);
sys = frd(sys, w)
From input 'input 1' to:
  Frequency(rad/s)
                           output 1
          10
                     0.293773+0.001033i
         100
                     0.294404+0.000109i
Continuous-time frequency response data.
sys = chgunits(sys,'Hz')
sys.freq
ans =
    1.5915
   15.9155
```

See Also

frd Create or convert to an FRD model get Get the properties of an LTI model set Set the properties of an LTI model **Purpose** Form a model with complex conjugate coefficients

Syntax sysc = conj(sys)

Description sysc = conj (sys) is an constructs a complex conjugate model sysc by applying

complex conjugation to all coefficients of the LTI model sys. This function accepts LTI models in transfer function (TF), zero/pole/gain (ZPK), and state

space (SS) formats.

Example If sys is the transfer function

(2+i)/(s+i)

then conj (sys) produces the transfer function

(2-i)/(s-i)

This operation is useful for manipulating partial fraction expansions.

See Also append Append LTI systems

Specify or convert to state-space form

tf Specify or convert to transfer function form zpk Specify or convert to zero-pole-gain form

Derive state-space model from block diagram description

Syntax

sysc = connect(sys,Q,inputs,outputs)

Description

Complex dynamical systems are often given in block diagram form. For systems of even moderate complexity, it can be quite difficult to find the state-space model required in order to bring certain analysis and design tools into use. Starting with a block diagram description, you can use append and connect to construct a state-space model of the system.

First, use

```
sys = append(sys1, sys2, ..., sysN)
```

to specify each block sysj in the diagram and form a block-diagonal, *unconnected* LTI model sys of the diagram.

Next, use

```
sysc = connect(sys,Q,inputs,outputs)
```

to connect the blocks together and derive a state-space model sysc for the overall interconnection. The arguments Q, inputs, and outputs have the following purpose:

- The matrix Q indicates how the blocks on the diagram are connected. It has a row for each input of sys, where the first element of each row is the input number. The subsequent elements of each row specify where the block input gets its summing inputs; negative elements indicate minus inputs to the summing junction. For example, if input 7 gets its inputs from the outputs 2, 15, and 6, where the input from output 15 is negative, the corresponding row of Q is [7 2 -15 6]. Short rows can be padded with trailing zeros (see example below).
- Given sys and Q, connect computes a state-space model of the interconnection with the same inputs and outputs as sys (that is, the concatenation of all block inputs and outputs). The index vectors inputs and outputs then indicate which of the inputs and outputs in the large unconnected system are external inputs and outputs of the block diagram. For example, if the external inputs are inputs 1, 2, and 15 of sys, and the external outputs are outputs 2 and 7 of sys, then inputs and outputs should be set to

```
inputs = [1 2 15];
outputs = [2 7];
```

The final model sysc has these particular inputs and outputs.

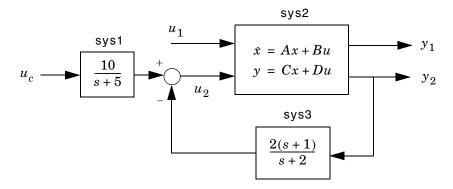
Since it is easy to make a mistake entering all the data required for a large model, be sure to verify your model in as many ways as you can. Here are some suggestions:

- Make sure the poles of the unconnected model sys match the poles of the various blocks in the diagram.
- Check that the final poles and DC gains are reasonable.
- Plot the step and bode responses of sysc and compare them with your expectations.

The connect function does not support delays in a reliable way. If you need to work extensively with block diagrams or you need to interconnect models with time delays, Simulink[®] is a much easier and more comprehensive tool for model building.

Example

Consider the following block diagram.



Given the matrices of the state-space model sys2,

```
A = [ -9.0201 17.7791
-1.6943 3.2138 ];
B = [ -.5112 .5362
```

```
-.002 -1.8470];

C = [ -3.2897 2.4544
-13.5009 18.0745];

D = [-.5476 -.1410
-.6459 .2958];
```

Define the three blocks as individual LTI models.

Next append these blocks to form the unconnected model sys.

```
sys = append(sys1, sys2, sys3)
```

This produces the block-diagonal model

sys

а	=	

	x1	x2	х3	x4
x1	- 5	0	0	0
x2	0	-9.0201	17.779	0
х3	0	-1.6943	3.2138	0
x4	0	0	0	-2

b	1	=

	uc	u1	u2	?
x1	4	0	0	0
x2	0	-0.5112	0.5362	0
х3	0	-0.002	-1.847	0
x4	0	0	0	1.4142

c =

	x1	x2	x3	x4
?	2.5	0	0	0
y1	0	-3.2897	2.4544	0
v2	0	-13.501	18.075	0

	?	0	0	0	-1.4142
d =					
		uc	u1	u2	?
	?	0	0	0	0
	y1	0	-0.5476	-0.141	0
	y2	0	-0.6459	0.2958	0
	?	0	0	0	2

Continuous-time system.

Note that the ordering of the inputs and outputs is the same as the block ordering you chose. Unnamed inputs or outputs are denoted b.

To derive the overall block diagram model from sys, specify the interconnections and the external inputs and outputs. You need to connect outputs 1 and 4 into input 3 (u2), and output 3 (y2) into input 4. The interconnection matrix Q is therefore

$$Q = [3 \ 1 \ -4 \ 4 \ 3 \ 0];$$

Note that the second row of Q has been padded with a trailing zero. The block diagram has two external inputs uc and u1 (inputs 1 and 2 of sys), and two external outputs y1 and y2 (outputs 2 and 3 of sys). Accordingly, set inputs and outputs as follows.

```
inputs = [1 2];
outputs = [2 3];
```

You can obtain a state-space model for the overall interconnection by typing

```
sysc = connect(sys,Q,inputs,outputs)
```

a =					
		x1	x2	x3	x4
	x1	-5	0	0	0
	x2	0.84223	0.076636	5.6007	0.47644
	x3	-2.9012	-33.029	45.164	-1.6411
	x4	0.65708	-11.996	16.06	-1.6283

connect

b =					
		uc	u1		
	x1	4	0		
	x2	0	-0.076001		
	х3	0	-1.5011		
	x4	0	-0.57391		
C =					
		x1	x2	х3	x4
	y1	-0.22148	-5.6818	5.6568	-0.12529
	y2	0.46463	-8.4826	11.356	0.26283
d =					
		uc	u1		
	y1	0	-0.66204		
	y2	0	-0.40582		

Continuous-time system.

Note that the inputs and outputs are as desired.

References

[1] Edwards, J.W., "A Fortran Program for the Analysis of Linear Continuous and Sampled-Data Systems," *NASA Report TM X56038*, Dryden Research Center, 1976.

See Also

append	Append LTI systems
feedback	Feedback connection
minreal	Minimal state-space realization
parallel	Parallel connection
series	Series connection

Output and state covariance of a system driven by white noise

Syntax

$$[P,Q] = covar(sys,W)$$

Description

covar calculates the stationary covariance of the output y of an LTI model sys driven by Gaussian white noise inputs w. This function handles both continuous- and discrete-time cases.

P = covar(sys,W) returns the steady-state output response covariance

$$P = E(yy^T)$$

given the noise intensity

$$\begin{split} E(w(t)w(\tau)^T) &= W \ \delta(t-\tau) \qquad \qquad \text{(continuous time)} \\ E(w[k]w[l]^T) &= W \ \delta_{kl} \qquad \qquad \text{(discrete time)} \end{split}$$

[P,Q] = covar(sys,W) also returns the steady-state state covariance

$$Q = E(xx^T)$$

when sys is a state-space model (otherwise ${\tt Q}$ is set to ${\tt []}$).

When applied to an N-dimensional LTI array sys, covar returns multidimensional arrays P, Q such that

P(:,:,i1,...iN) and Q(:,:,i1,...iN) are the covariance matrices for the model sys(:,:,i1,...iN).

Example

Compute the output response covariance of the discrete SISO system

$$H(z) = \frac{2z+1}{z^2+0.2z+0.5}$$
, $T_s = 0.1$

due to Gaussian white noise of intensity W = 5. Type

and MATLAB® returns

You can compare this output of covar to simulation results.

```
randn('seed',0)
w = sqrt(5)*randn(1,1000); % 1000 samples
% Simulate response to w with LSIM:
y = lsim(sys,w);
% Compute covariance of y values
psim = sum(y .* y)/length(w);
This yields
psim =
```

The two covariance values p and psim do not agree perfectly due to the finite simulation horizon.

Algorithm

Transfer functions and zero-pole-gain models are first converted to state space with ss.

For continuous-time state-space models

$$\dot{x} = Ax + Bw$$
$$y = Cx + Dw$$

32,6269

 ${\cal Q}$ is obtained by solving the Lyapunov equation

$$AQ + QA^T + BWB^T = 0$$

The output response covariance P is finite only when D=0 and then $P=CQC^T$.

In discrete time, the state covariance solves the discrete Lyapunov equation

$$AQA^{T} - Q + BWB^{T} = 0$$

and P is given by $P = CQC^{T} + DWD^{T}$

Note that P is well defined for nonzero D in the discrete case.

Limitations The state and output covariances are defined for *stable* systems only. For

continuous systems, the output response covariance P is finite only when the

D matrix is zero (strictly proper system).

References [1] Bryson, A.E. and Y.C. Ho, *Applied Optimal Control*, Hemisphere

Publishing, 1975, pp. 458-459.

See Also dlyap Solver for discrete-time Lyapunov equations

1yap Solver for continuous-time Lyapunov equations

Form the controllability matrix

Syntax

Description

ctrb computes the controllability matrix for state-space systems. For an n-by-n matrix A and an n-by-m matrix B, ctrb (A,B) returns the controllability matrix

$$Co = \left[B AB A^2 B \dots A^{n-1} B \right]$$
 (1-1)

where Co has n rows and nm columns.

Co = ctrb(sys) calculates the controllability matrix of the state-space LTI object sys. This syntax is equivalent to executing

$$Co = ctrb(sys.A, sys.B)$$

The system is controllable if Co has full rank n.

Example

Check if the system with the following data

is controllable. Type

% Number of uncontrollable states unco=length(A)-rank(Co)

and MATLAB returns

Limitations

Estimating the rank of the controllability matrix is ill-conditioned; that is, it is very sensitive to roundoff errors and errors in the data. An indication of this can be seen from this simple example.

$$A = \begin{bmatrix} 1 & \delta \\ 0 & 1 \end{bmatrix}, \qquad B = \begin{bmatrix} 1 \\ \delta \end{bmatrix}$$

This pair is controllable if $\delta \neq 0$ but if $\delta < \sqrt{eps}$, where eps is the relative machine precision. ctrb(A,B) returns

$$\begin{bmatrix} B & AB \end{bmatrix} = \begin{bmatrix} 1 & 1 \\ \delta & \delta \end{bmatrix}$$

which is not full rank. For cases like these, it is better to determine the controllability of a system using ctrbf.

See Also

ctrbf obsv Compute the controllability staircase form Compute the observability matrix

Compute the controllability staircase form

Syntax

Description

If the controllability matrix of (A,B) has rank $r \le n$, where n is the size of A, then there exists a similarity transformation such that

$$\overline{A} = TAT^T$$
, $\overline{B} = TB$, $\overline{C} = CT^T$

where *T* is unitary, and the transformed system has a *staircase* form, in which the uncontrollable modes, if there are any, are in the upper left corner.

$$\overline{A} = \begin{bmatrix} A_{uc} & 0 \\ A_{21} & A_c \end{bmatrix}, \qquad \overline{B} = \begin{bmatrix} 0 \\ B_c \end{bmatrix}, \qquad \overline{C} = \begin{bmatrix} C_{nc} & C_c \end{bmatrix}$$

where (A_c,B_c) is controllable, all eigenvalues of A_{uc} are uncontrollable, and

$$C_c(sI - A_c)^{-1}B_c = C(sI - A)^{-1}B.$$

[Abar,Bbar,Cbar,T,k] = ctrbf(A,B,C) decomposes the state-space system represented by A, B, and C into the controllability staircase form, Abar, Bbar, and Cbar, described above. T is the similarity transformation matrix and k is a vector of length n, where n is the order of the system represented by A. Each entry of k represents the number of controllable states factored out during each step of the transformation matrix calculation. The number of nonzero elements in k indicates how many iterations were necessary to calculate T, and sum(k) is the number of states in A_c , the controllable portion of Abar.

ctrbf(A,B,C,to1) uses the tolerance to1 when calculating the controllable/uncontrollable subspaces. When the tolerance is not specified, it defaults to 10*n*norm(A,1)*eps.

Example

Compute the controllability staircase form for

1

0

and locate the uncontrollable mode.

[Abar,Bbar,Cbar,T,k]=ctrbf(A,B,C)

The decomposed system Abar shows an uncontrollable mode located at -3 and a controllable mode located at 2.

Algorithm

 ${\tt ctrbf}$ is an M-file that implements the Staircase Algorithm of [1].

References

[1] Rosenbrock, M.M., State-Space and Multivariable Theory, John Wiley, 1970.

See Also

ctrb Form the controllability matrix
minreal Minimum realization and pole-zero cancellation

Convert discrete-time LTI models to continuous time

Syntax

```
sysc = d2c(sysd)
```

$$sysc = d2c(sysd, method)$$

Description

d2c converts LTI models from discrete to continuous time using one of the following conversion methods:

 $\mbox{'zoh'}$ Zero-order hold on the inputs. The control inputs are

assumed piecewise constant over the sampling period.

'tustin' Bilinear (Tustin) approximation to the derivative.

'prewarp' Tustin approximation with frequency prewarping.

'matched' Matched pole-zero method of [1] (for SISO systems only).

The string *method* specifies the conversion method. If *method* is omitted then zero-order hold ('zoh') is assumed. See "Continuous/Discrete Conversions of LTI Models" for more details on the conversion methods.

Example

Consider the discrete-time model with transfer function

$$H(z) = \frac{z - 1}{z^2 + z + 0.3}$$

and sample time $T_s = 0.1$ second. You can derive a continuous-time zero-order-hold equivalent model by typing

$$Hc = d2c(H)$$

Discretizing the resulting model Hc with the zero-order hold method (this is the default method) and sampling period $T_{\rm s}$ = 0.1 gives back the original discrete model H(z). To see this, type

To use the Tustin approximation instead of zero-order hold, type

$$Hc = d2c(H, 'tustin')$$

As with zero-order hold, the inverse discretization operation

```
c2d(Hc,0.1, 'tustin')
```

gives back the original H(z).

Algorithm

The 'zoh' conversion is performed in state space and relies on the matrix logarithm (see logm in the MATLAB documentation).

Limitations

The Tustin approximation is not defined for systems with poles at z=-1 and is ill-conditioned for systems with poles near z=-1.

The zero-order hold method cannot handle systems with poles at z=0. In addition, the 'zoh' conversion increases the model order for systems with negative real poles, [2]. This is necessary because the matrix logarithm maps real negative poles to complex poles. As a result, a discrete model with a single pole at z=-0.5 would be transformed to a continuous model with a single complex pole at $log(-0.5)\approx -0.6931+j\pi$. Such a model is not meaningful because of its complex time response.

To ensure that all complex poles of the continuous model come in conjugate pairs, d2c replaces negative real poles $z=-\alpha$ with a pair of complex conjugate poles near $-\alpha$. The conversion then yields a continuous model with higher order. For example, the discrete model with transfer function

$$H(z) = \frac{z + 0.2}{(z + 0.5)(z^2 + z + 0.4)}$$

and sample time 0.1 second is converted by typing

Ts = 0.1
H =
$$zpk(-0.2, -0.5, 1, Ts) * tf(1,[1 1 0.4], Ts)$$

Hc = $d2c(H)$

MATLAB responds with

Warning: System order was increased to handle real negative poles.

Convert Hc back to discrete time by typing

This discrete model coincides with H(z) after canceling the pole/zero pair at z=-0.5.

References

[1] Franklin, G.F., J.D. Powell, and M.L. Workman, *Digital Control of Dynamic Systems*, Second Edition, Addison-Wesley, 1990.

[2] Kollár, I., G.F. Franklin, and R. Pintelon, "On the Equivalence of z-domain and s-domain Models in System Identification," *Proceedings of the IEEE Instrumentation and Measurement Technology Conference*, Brussels, Belgium, June, 1996, Vol. 1, pp. 14-19.

See Also

c2d	Continuous- to discrete-time conversion
d2d	Resampling of discrete models
logm	Matrix logarithm

Resample discrete-time LTI models or add input delays

Syntax

$$sys1 = d2d(sys,Ts)$$

Description

sys1 = d2d(sys,Ts) resamples the discrete-time LTI model sys to produce an equivalent discrete-time model sys1 with the new sample time Ts (in seconds). The resampling assumes zero-order hold on the inputs and is equivalent to consecutive d2c and c2d conversions.

$$sys1 = c2d(d2c(sys),Ts)$$

Example

Consider the zero-pole-gain model

$$H(z) = \frac{z - 0.7}{z - 0.5}$$

with sample time 0.1 second. You can resample this model at 0.05 second by typing

```
H = zpk(0.7,0.5,1,0.1)
H2 = d2d(H,0.05)

Zero/pole/gain:
(z-0.8243)
------
(z-0.7071)

Sampling time: 0.05
```

Note that the inverse resampling operation, performed by typing d2d(H2,0.1), yields back the initial model H(z).

```
Zero/pole/gain:
(z-0.7)
-----
(z-0.5)

Sampling time: 0.1
```

See Also

Continuous- to discrete-time conversion d2c Discrete- to continuous-time conversion

Compute damping factors and natural frequencies

Syntax

Description

damp calculates the damping factor and natural frequencies of the poles of an LTI model sys. When invoked without lefthand arguments, a table of the eigenvalues in increasing frequency, along with their damping factors and natural frequencies, is displayed on the screen.

[Wn,Z] = damp(sys) returns column vectors Wn and Z containing the natural frequencies ω_n and damping factors ζ of the poles of sys. For discrete-time systems with poles z and sample time T_s , damp computes "equivalent" continuous-time poles s by solving

$$z = e^{sT_s}$$

The values \mbox{Wn} and \mbox{Z} are then relative to the continuous-time poles s. Both \mbox{Wn} and \mbox{Z} are empty if the sample time is unspecified.

[Wn,Z,P] = damp(sys) returns an additional vector P containing the (true) poles of sys. Note that P returns the same values as pole(sys) (up to reordering).

Example

Compute and display the eigenvalues, natural frequencies, and damping factors of the continuous transfer function

$$H(s) = \frac{2s^2 + 5s + 1}{s^2 + 2s + 3}$$

Type

$$H = tf([2 5 1],[1 2 3])$$

Transfer function: 2 s^2 + 5 s + 1s^2 + 2 s + 3

Type

damp

damp(H)

and MATLAB returns

Eigenvalue

-1.00e+000 + 1.41e+000i	5.77e-001	1.73e+000
-1.00e+000 - 1.41e+000i	5.77e-001	1.73e+000

Damping

Freq. (rad/s)

See Also

eig Calculate eigenvalues and eigenvectors

esort,dsort Sort system poles
pole Compute system poles
pzmap Pole-zero map

zero Compute (transmission) zeros

Solve discrete-time algebraic Riccati equations (DARE)

Syntax

Description

[X,L,G] = dare(A,B,Q,R) computes the unique stabilizing solution X of the discrete-time algebraic Riccati equation

$$A^{T}XA - X - A^{T}XB(B^{T}XB + R)^{-1}B^{T}XA + Q = 0$$

The dare function also returns the gain matrix, $G = (B^T X B + R)^{-1} B^T X A$, and the vector L of closed loop eigenvalues, where

[X,L,G] = dare(A,B,Q,R,S,E) solves the more general discrete-time algebraic Riccati equation,

$$A^{T}XA - E^{T}XE - (A^{T}XB + S)(B^{T}XB + R)^{-1}(B^{T}XA + S^{T}) + Q = 0$$

or, equivalently, if R is nonsingular,

$$E^{T}XE = F^{T}XF + -F^{T}XB(B^{T}XB + R)^{-1}B^{T}XF + Q - SR^{-1}S^{T}$$

where $F = A - BR^{-1}S$. When omitted, R, S, and E are set to the default values R=I, S=0, and E=I.

The dare function returns the corresponding gain matrix

$$G = (B^T X B + R)^{-1} (B^T X A + S^T)$$

and a vector ${\sf L}$ of closed-loop eigenvalues, where

[X,L,G,report] = dare(A,B,Q,...) returns a diagnosis report with value:

 -1 when the associated symplectic pencil has eigenvalues on or very near the unit circle

- -2 when there is no finite stabilizing solution X
- The Frobenius norm if X exists and is finite

[X1,X2,L,report] = dare(A,B,Q,...,'factor') returns two matrices, X1 and X2, and a diagonal scaling matrix D such that X = D*(X2/X1)*D. The vector L contains the closed-loop eigenvalues. All outputs are empty when the associated Symplectic matrix has eigenvalues on the unit circle.

Algorithm

dare implements the algorithms described in [1]. It uses the QZ algorithm to deflate the extended symplectic pencil and compute its stable invariant subspace.

Limitations

The (A,B) pair must be stabilizable (that is, all eigenvalues of A outside the unit disk must be controllable). In addition, the associated symplectic pencil must have no eigenvalue on the unit circle. Sufficient conditions for this to hold are (Q,A) detectable when S=0 and R>0, or

$$\begin{bmatrix} Q & S \\ S^T & R \end{bmatrix} > 0$$

References

[1] Arnold, W.F., III and A.J. Laub, "Generalized Eigenproblem Algorithms and Software for Algebraic Riccati Equations," *Proc. IEEE*, 72 (1984), pp. 1746-1754.

See Also

care	Solve continuous-time algebraic Riccati equations
dlyap	Solve discrete-time Lyapunov equations
gdare	Solve generalized discrete-time algebraic Riccati
	equations

dcgain

Purpose

Compute low frequency (DC) gain of LTI system

Syntax

k = dcgain(sys)

Description

k = dcgain(sys) computes the DC gain k of the LTI model sys.

Continuous Time

The continuous-time DC gain is the transfer function value at the frequency s = 0. For state-space models with matrices (A, B, C, D), this value is

$$K = D - CA^{-1}B$$

Discrete Time

The discrete-time DC gain is the transfer function value at z=1. For state-space models with matrices (A,B,C,D), this value is

$$K = D + C(I - A)^{-1}B$$

Remark

The DC gain is infinite for systems with integrators.

Example

To compute the DC gain of the MIMO transfer function

$$H(s) = \begin{bmatrix} 1 & \frac{s-1}{s^2 + s + 3} \\ \frac{1}{s+1} & \frac{s+2}{s-3} \end{bmatrix}$$

type

$$\label{eq:Hamiltonian} \begin{array}{lll} H = & [1 \ tf([1 \ -1],[1 \ 1 \ 3]) \ ; \ tf(1,[1 \ 1]) \ tf([1 \ 2],[1 \ -3])] \\ dcgain(H) \end{array}$$

See Also

evalfr norm Evaluates frequency response at single frequency LTI system norms

Replace delays of discrete-time TF, SS, or ZPK models by poles at z=0, or replace delays of FRD models by a phase shift

Syntax

sys = delay2z(sys)

Description

sys = delay2z(sys) maps all time delays to poles at z=0 for discrete-time TF, ZPK, or SS models sys. Specifically, a delay of k sampling periods is replaced by (1/z)^k in the transfer function corresponding to the model.

For FRD models, delay2z absorbs all time delays into the frequency response data, and is applicable to both continuous- and discrete-time FRDs.

Example

See Also

hasdelay True for LTI models with delays
pade Pade approximation of time delays
totaldelay Combine delays for an LTI model

dlqr

Purpose

Design linear-quadratic (LQ) state-feedback regulator for discrete-time plant

Syntax

$$[K,S,e] = dlqr(a,b,Q,R)$$

 $[K,S,e] = dlqr(a,b,Q,R,N)$

Description

[K,S,e] = dlqr(a,b,Q,R,N) calculates the optimal gain matrix K such that the state-feedback law

$$u[n] = -Kx[n]$$

minimizes the quadratic cost function

$$J(u) = \sum_{n=1}^{\infty} (x[n]^{T}Qx[n] + u[n]^{T}Ru[n] + 2x[n]^{T}Nu[n])$$

for the discrete-time state-space mode

$$lx[n+1] = Ax[n] + Bu[n]$$

The default value N=0 is assumed when N is omitted.

In addition to the state-feedback gain K, $\tt dlqr$ returns the infinite horizon solution S of the associated discrete-time Riccati equation

$$A^{T}SA - S - (A^{T}SB + N)(B^{T}SB + R)^{-1}(B^{T}SA + N^{T}) + Q = 0$$

and the closed-loop eigenvalues e = eig(a-b*K). Note that K is derived from S by

$$K = (B^T S B + R)^{-1} (B^T S A + N^T)$$

Limitations

The problem data must satisfy:

- The pair (A, B) is stabilizable.
- R > 0 and $Q NR^{-1}N^T \ge 0$.
- $(Q NR^{-1}N^T, A BR^{-1}N^T)$ has no unobservable mode on the unit circle.

See Also

dare Solve discrete Riccati equations

lqgreg LQG regulator

dlqr

lqr	State-feedback LQ regulator for continuous plant
lqrd	Discrete LQ regulator for continuous plant
lqry	State-feedback LQ regulator with output weighting

dlyap

Purpose

Solve discrete-time Lyapunov equations

Syntax

X = dlyap(A,Q)
X = dlyap(A,B,C)
X = dlyap(A,Q,[],E)

Description

X = dlyap(A,Q) solves the discrete-time Lyapunov equation

$$AXA^T - X + Q = 0$$

where A and Q are n-by-n matrices.

The solution X is symmetric when Q is symmetric, and positive definite when Q is positive definite and A has all its eigenvalues inside the unit disk.

X = dlyap(A,B,C) solves the Sylvester equation

$$AXB^T - X + C = 0$$

where A, B, and C must have compatible dimensions but need not be square.

X = dlyap(A,Q,[],E) solves the generalized discrete-time Lyapunov equation

$$AXA^T - EXE^T + Q = 0$$

where Q is a symmetric matrix. The empty square brackets, [], are mandatory. If you place any values inside them, the function will error out.

Algorithm

dlyap uses SLICOT routines SB03MD and SG03AD for Lyapunov equations and SB04QD (SLICOT) for Sylvester equations.

Diagnostics

The discrete-time Lyapunov equation has a (unique) solution if the eigenvalues $\alpha_1, \alpha_2, ..., \alpha_n$ of A satisfy $\alpha_i \alpha_j \neq 1$ for all (i,j).

If this condition is violated, dlyap produces the error message

Solution does not exist or is not unique.

See Also

covar Covariance of system response to white noise 1yap Solve continuous Lyapunov equations **Purpose** Square-root solver for continuous-time Lyapunov equations

Syntax R = dlyapchol(A,B)

R = dlyapchol(A,B,E)

Description R = dlyapchol(A,B) computes a Cholesky factorization X = R'*R of the

solution X to the Lyapunov matrix equation:

A*X*A' - X + B*B' = 0

All eigenvalues of A matrix must lie in the open unit disk for R to exist.

X = dlyapchol(A,B,E) computes a Cholesky factorization X = R'*R of X solving

the Sylvester equation

A*X*A' - E*X*E' + B*B' = 0

All generalized eigenvalues of (A,E) must lie in the open unit disk for R to exist.

Algorithm dlyapchol uses SLICOT routines SB03OD and SG03BD.

See Also Solver for discrete-time Lyapunov equations

lyapchol Square-root solver for continuous-time Lyapunov

equations

Generate stable random discrete test models

Syntax

```
sys = drss(n)
sys = drss(n,p)
sys = drss(n,p,m)
sys = drss(n,p,m,s1,...sn)
```

Description

sys = drss(n) produces a random n-th order stable model with one input and one output, and returns the model in the state-space object sys.

drss(n,p) produces a random n-th order stable model with one input and p outputs.

drss(n,m,p) generates a random n-th order stable model with m inputs and p outputs.

drss(n,p,m,s1,...sn) generates a s1-by-sn array of random n-th order stable model with m inputs and p outputs.

In all cases, the discrete-time state-space model or array returned by drss has an unspecified sampling time. To generate transfer function or zero-pole-gain systems, convert sys using tf or zpk.

Example

Generate a random discrete LTI system with three states, two inputs, and two outputs.

sys = di	rss(3,2,2	2)		
a =				
		x1	x2	х3
	x1	0.38630	-0.21458	-0.09914
	x2	-0.23390	-0.15220	-0.06572
	х3	-0.03412	0.11394	-0.22618
b =				
		u1	u2	
	x1	0.98833	0.51551	
	x2	0	0.33395	
	х3	0.42350	0.43291	

c =				
		x1	x2	x3
	y1	0.22595	0.76037	0
	y2	0	0	0
d =				
		u1	u2	
	y1	0	0.68085	
	y2	0.78333	0.46110	

Sampling time: unspecified Discrete-time system.

See Also

rss	Generate stable random continuous test models
tf	Convert LTI systems to transfer functions form
zpk	Convert LTI systems to zero-pole-gain form

dsort

Purpose

Sort discrete-time poles by magnitude

Syntax

```
s = dsort(p)
[s,ndx] = dsort(p)
```

Description

dsort sorts the discrete-time poles contained in the vector p in descending order by magnitude. Unstable poles appear first.

When called with one lefthand argument, dsort returns the sorted poles in s.

[s,ndx] = dsort(p) also returns the vector ndx containing the indices used in the sort.

Example

Sort the following discrete poles.

Limitations

The poles in the vector p must appear in complex conjugate pairs.

See Also

eig Calculate eigenvalues and eigenvectors

esort, sort Sort system poles
pole Compute system poles

pzmap Pole-zero map

zero Compute (transmission) zeros

Specify descriptor state-space models

Syntax

```
sys = dss(a,b,c,d,e)
sys = dss(a,b,c,d,e,Ts)
sys = dss(a,b,c,d,e,ltisys)

sys = dss(a,b,c,d,e,'Property1',Value1,...,'PropertyN',ValueN)
sys = dss(a,b,c,d,e,Ts,'Property1',Value1,...,'PropertyN',ValueN)
```

Description

sys = dss(a,b,c,d,e) creates the continuous-time descriptor state-space model

$$E\dot{x} = Ax + Bu$$
$$v = Cx + Du$$

The E matrix must be nonsingular. The output sys is an SS model storing the model data (see "LTI Objects" on page 2-3). Note that ss produces the same type of object. If the matrix D=0, do can simply set d to the scalar 0 (zero).

sys = dss(a,b,c,d,e,Ts) creates the discrete-time descriptor model

$$Ex[n+1] = Ax[n] + Bu[n]$$
$$y[n] = Cx[n] + Du[n]$$

with sample time Ts (in seconds).

sys = dss(a,b,c,d,e,ltisys) creates a descriptor model with generic LTI properties inherited from the LTI model ltisys (including the sample time). See "LTI Properties" on page 2-26 for an overview of generic LTI properties.

Any of the previous syntaxes can be followed by property name/property value pairs

```
'Property', Value
```

Each pair specifies a particular LTI property of the model, for example, the input names or some notes on the model history. See set and the example below for details.

Example

The command

creates the model

$$5\dot{x} = x + 2u$$
$$y = 3x + 4u$$

with a 0.1 second input delay. The input is labeled 'voltage', and a note is attached to tell you that this is just an example.

See Also

dssdata	Retrieve A, B, C, D, E matrices of descriptor model
get	Get properties of LTI models
set	Set properties of LTI models

Specify (regular) state-space models SS

Purpose Quick access to descriptor state-space data

Syntax [a,b,c,d,e] = dssdata(sys)

[a,b,c,d,e,Ts] = dssdata(sys)

Description [a,b,c,d,e] = dssdata(sys) extracts the descriptor matrix data

(A,B,C,D,E) from the state-space model sys. If sys is a transfer function or zero-pole-gain model, it is first converted to state space. Note that dssdata is

then equivalent to ssdata because it always returns E = I.

[a,b,c,d,e,Ts] = dssdata(sys) also returns the sample time Ts.

You can access the remaining LTI properties of sys with get or by direct

referencing, for example,

sys.notes

See Also dss Specify descriptor state-space models

get Get properties of LTI models ssdata Quick access to state-space data

tfdata Quick access to transfer function data zpkdata Quick access to zero-pole-gain data

Sort continuous-time poles by real part

Syntax

```
s = esort(p)
[s,ndx] = esort(p)
```

Description

esort sorts the continuous-time poles contained in the vector p by real part. Unstable eigenvalues appear first and the remaining poles are ordered by decreasing real parts.

When called with one left-hand argument, s = esort(p) returns the sorted eigenvalues in s.

[s,ndx] = esort(p) returns the additional argument ndx, a vector containing the indices used in the sort.

Example

Sort the following continuous eigenvalues.

```
p = 
    -0.2410+ 0.5573i
    -0.2410- 0.5573i
    0.1503
    -0.0972
    -0.2590

esort(p)

ans = 
    0.1503
    -0.0972
    -0.2410+ 0.5573i
    -0.2410- 0.5573i
    -0.2590
```

Limitations

The eigenvalues in the vector p must appear in complex conjugate pairs.

See Also

dsort, sort Sort system poles

eig Calculate eigenvalues and eigenvectors

pole Compute system poles

pzmap Pole-zero map

zero

 $Compute\ (transmission)\ zeros$

Form state estimator given estimator gain

Syntax

Description

est = estim(sys,L) produces a state/output estimator est given the plant state-space model sys and the estimator gain L. All inputs w of sys are assumed stochastic (process and/or measurement noise), and all outputs y are measured. The estimator est is returned in state-space form (SS object). For a continuous-time plant sys with equations

$$\dot{x} = Ax + Bw$$
$$y = Cx + Dw$$

estim generates plant output and state estimates \hat{y} and \hat{x} as given by the following model.

$$\hat{x} = A\hat{x} + L(y - C\hat{x})$$

$$\begin{bmatrix} \hat{y} \\ \hat{x} \end{bmatrix} = \begin{bmatrix} C \\ I \end{bmatrix} \hat{x}$$

The discrete-time estimator has similar equations.

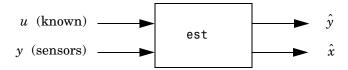
est = estim(sys,L,sensors,known) handles more general plants sys with both known inputs u and stochastic inputs w, and both measured outputs y and nonmeasured outputs z.

$$\begin{split} \dot{x} &= Ax + B_1 w + B_2 u \\ \begin{bmatrix} z \\ y \end{bmatrix} &= \begin{bmatrix} C_1 \\ C_2 \end{bmatrix} x + \begin{bmatrix} D_{11} \\ D_{21} \end{bmatrix} w + \begin{bmatrix} D_{12} \\ D_{22} \end{bmatrix} u \end{split}$$

The index vectors sensors and known specify which outputs y are measured and which inputs u are known. The resulting estimator est uses both u and y to produce the output and state estimates.

$$\hat{x} = A\hat{x} + B_2 u + L(y - C_2 \hat{x} - D_{22} u)$$

$$\begin{bmatrix} \hat{y} \\ \hat{x} \end{bmatrix} = \begin{bmatrix} C_2 \\ I \end{bmatrix} \hat{x} + \begin{bmatrix} D_{22} \\ 0 \end{bmatrix} u$$



estim handles both continuous- and discrete-time cases. You can use the functions place (pole placement) or kalman (Kalman filtering) to design an adequate estimator gain L. Note that the estimator poles (eigenvalues of A-LC) should be faster than the plant dynamics (eigenvalues of A) to ensure accurate estimation.

Example

Consider a state-space model sys with seven outputs and four inputs. Suppose you designed a Kalman gain matrix L using outputs 4, 7, and 1 of the plant as sensor measurements, and inputs 1,4, and 3 of the plant as known (deterministic) inputs. You can then form the Kalman estimator by

```
sensors = [4,7,1];
known = [1,4,3];
est = estim(sys,L,sensors,known)
```

See the function kalman for direct Kalman estimator design.

See Also

kalman	Design Kalman estimator
place	Pole placement
reg	Form regulator given state-feedback and estimator
	gains

evalfr

Purpose

Evaluate frequency response at a single (complex) frequency

Syntax

frsp = evalfr(sys,f)

Description

frsp = evalfr(sys,f) evaluates the transfer function of the TF, SS, or ZPK model sys at the complex number f. For state-space models with data (A,B,C,D), the result is

$$H(f) = D + C(fI - A)^{-1}B$$

evalfr is a simplified version of freqresp meant for quick evaluation of the response at a single point. Use freqresp to compute the frequency response over a set of frequencies.

Example

To evaluate the discrete-time transfer function

$$H(z) = \frac{z-1}{z^2+z+1}$$
 at $z = 1+j$, type
$$H = tf([1 -1],[1 1 1],-1)$$

$$z = 1+j$$
 evalfr(H,z) ans = 2.3077e-01 + 1.5385e-01i

Limitations

The response is not finite when f is a pole of sys.

See Also

bode Bode frequency response frequesp Frequency response over a set of frequencies

rrequency response over a set of frequency

sigma Singular value response

Feedback connection of two LTI models

Syntax

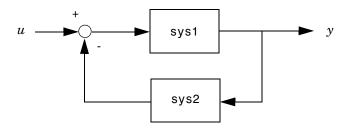
```
sys = feedback(sys1,sys2)
```

sys = feedback(sys1,sys2,sign)

sys = feedback(sys1,sys2,feedin,feedout,sign)

Description

sys = feedback(sys1,sys2) returns an LTI model sys for the negative feedback interconnection.



The closed-loop model sys has u as input vector and y as output vector. The LTI models sys1 and sys2 must be both continuous or both discrete with identical sample times. Precedence rules are used to determine the resulting model type (see Precedence Rules).

To apply positive feedback, use the syntax

```
sys = feedback(sys1, sys2, +1)
```

By default, feedback(sys1,sys2) assumes negative feedback and is equivalent to feedback(sys1,sys2,-1).

Finally,

```
sys = feedback(sys1,sys2,feedin,feedout)
```

v + sys1 y y

sys

computes a closed-loop model sys for the more general feedback loop.

The vector feedin contains indices into the input vector of sys1 and specifies which inputs u are involved in the feedback loop. Similarly, feedout specifies which outputs y of sys1 are used for feedback. The resulting LTI model sys has the same inputs and outputs as sys1 (with their order preserved). As before, negative feedback is applied by default and you must use

to apply positive feedback.

For more complicated feedback structures, use append and connect.

Remark

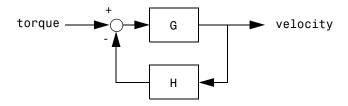
You can specify static gains as regular matrices, for example,

$$sys = feedback(sys1,2)$$

However, at least one of the two arguments sys1 and sys2 should be an LTI object. For feedback loops involving two static gains k1 and k2, use the syntax

$$sys = feedback(tf(k1),k2)$$

Examples Example 1



To connect the plant

$$G(s) = \frac{2s^2 + 5s + 1}{s^2 + 2s + 3}$$

with the controller

$$H(s) = \frac{5(s+2)}{s+10}$$

using negative feedback, type

and MATLAB returns

```
Zero/pole/gain from input "torque" to output "velocity": 0.18182 (s+10) (s+2.281) (s+0.2192) (s+3.419) (s^2 + 1.763s + 1.064)
```

The result is a zero-pole-gain model as expected from the precedence rules. Note that Cloop inherited the input and output names from G.

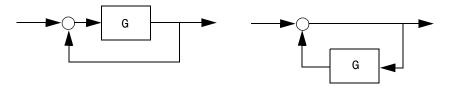
Example 2

Consider a state-space plant P with five inputs and four outputs and a state-space feedback controller K with three inputs and two outputs. To connect outputs 1, 3, and 4 of the plant to the controller inputs, and the controller outputs to inputs 4 and 2 of the plant, use

```
feedin = [4 2];
feedout = [1 3 4];
Cloop = feedback(P,K,feedin,feedout)
```

Example 3

You can form the following negative-feedback loops



by

```
Cloop = feedback(G,1) % left diagram
Cloop = feedback(1,G) % right diagram
```

Limitations

The feedback connection should be free of algebraic loop. If D_1 and D_2 are the feedthrough matrices of sys1 and sys2, this condition is equivalent to:

- $I + D_1D_2$ nonsingular when using negative feedback
- $I D_1D_2$ nonsingular when using positive feedback.

See Also

series	Series connection
parallel	Parallel connection

connect Derive state-space model for block diagram

interconnection

Specify discrete transfer functions in DSP format

Syntax

```
sys = filt(num,den)
sys = filt(num,den,Ts)
sys = filt(M)

sys = filt(num,den,'Property1',Value1,...,'PropertyN',ValueN)
sys = filt(num,den,Ts,'Property1',Value1,...,'PropertyN',ValueN)
```

Description

In digital signal processing (DSP), it is customary to write transfer functions as rational expressions in z^{-1} and to order the numerator and denominator terms in *ascending* powers of z^{-1} , for example,

$$H(z^{-1}) = \frac{2 + z^{-1}}{1 + 0.4z^{-1} + 2z^{-2}}$$

The function filt is provided to facilitate the specification of transfer functions in DSP format.

sys = filt(num,den) creates a discrete-time transfer function sys with
numerator(s) num and denominator(s) den. The sample time is left unspecified
(sys.Ts = -1) and the output sys is a TF object.

sys = filt(num,den,Ts) further specifies the sample time Ts (in seconds).

sys = filt(M) specifies a static filter with gain matrix M.

Any of the previous syntaxes can be followed by property name/property value pairs of the form

'Property', Value

Each pair specifies a particular LTI property of the model, for example, the input names or the transfer function variable. See LTI Properties and the set entry for additional information on LTI properties and admissible property values.

Arguments

For SISO transfer functions, num and den are row vectors containing the numerator and denominator coefficients ordered in ascending powers of z^{-1} . For example, den = [1 0.4 2] represents the polynomial $1 + 0.4z^{-1} + 2z^{-2}$.

MIMO transfer functions are regarded as arrays of SISO transfer functions (one per I/O channel), each of which is characterized by its numerator and denominator. The input arguments num and den are then cell arrays of row vectors such that:

- num and den have as many rows as outputs and as many columns as inputs.
- Their (i,j) entries $num\{i,j\}$ and $den\{i,j\}$ specify the numerator and denominator of the transfer function from input j to output i.

If all SISO entries have the same denominator, you can also set den to the row vector representation of this common denominator. See also MIMO Transfer Function Models for alternative ways to specify MIMO transfer functions.

Remark

filt behaves as tf with the Variable property set to $'z^-1'$ or 'q'. See tf entry below for details.

Example

Typing the commands

```
num = {1 , [1 0.3]}
den = {[1 1 2] ,[5 2]}
H = filt(num,den,'inputname',{'channel1' 'channel2'})
```

creates the two-input digital filter

$$H(z^{-1}) = \begin{bmatrix} \frac{1}{1+z^{-1}+2z^{-2}} & \frac{1+0.3z^{-1}}{5+2z^{-1}} \end{bmatrix}$$

with unspecified sample time and input names 'channel1' and 'channel2'.

See Also

tf	Create transfer functions
zpk	Create zero-pole-gain models
SS	Create state-space models

Create a frequency response data (FRD) object or convert another model type to an FRD model

Syntax

```
sys = frd(response,frequency)
sys = frd(response,frequency,Ts)
sys = frd
sys = frd(response,frequency,ltisys)

sysfrd = frd(sys,frequency)
sysfrd = frd(sys,frequency,'Units',units)
```

Description

sys = frd(response, frequency) creates an FRD model sys from the frequency response data stored in the multidimensional array response. The vector frequency represents the underlying frequencies for the frequency response data. See Table 1-1, Data Format for the Argument response in FRD Models.

sys = frd(response, frequency, Ts) creates a discrete-time FRD model sys with scalar sample time Ts. Set Ts = -1 to create a discrete-time FRD model without specifying the sample time.

sys = frd creates an empty FRD model.

The input argument list for any of these syntaxes can be followed by property name/property value pairs of the form

```
'PropertyName',PropertyValue
```

You can use these extra arguments to set the various properties of FRD models (see the set command, or LTI Properties and Model-Specific Properties). These properties include 'Units'. The default units for FRD models are in 'rad/s'.

To force an FRD model sys to inherit all of its generic LTI properties from any existing LTI model refsys, use the syntax

```
sys = frd(response, frequency, ltisys)
```

sysfrd = frd(sys,frequency) converts a TF, SS, or ZPK model to an FRD model. The frequency response is computed at the frequencies provided by the vector frequency.

sysfrd = frd(sys,frequency,'Units',units)converts an FRD model from a TF, SS, or ZPK model while specifying the units for frequency to be units ('rad/s' or 'Hz').

Arguments

When you specify a SISO or MIMO FRD model, or an array of FRD models, the input argument frequency is always a vector of length Nf, where Nf is the number of frequency data points in the FRD. The specification of the input argument response is summarized in the following table.

Table 1-1: Data Format for the Argument response in FRD Models

Model Form	Response Data Format
SISO model	Vector of length Nf for which response(i) is the frequency response at the frequency frequency(i)
MIMO model with Ny outputs and Nu inputs	Ny-by-Nu-by-Nf multidimensional array for which response (i,j,k) specifies the frequency response from input j to output i at frequency frequency(k)
S1-byby-Sn array of models with Ny outputs and Nu inputs	Multidimensional array of size [Ny Nu S1 Sn] for which response(i,j,k,:) specifies the array of frequency response data from input j to output i at frequency frequency(k)

Remarks

See Frequency Response Data (FRD) Models for more information on single FRD models, and Creating LTI Models for information on building arrays of FRD models.

Example

Type the commands

```
freq = logspace(1,2);
resp = .05*(freq).*exp(i*2*freq);
sys = frd(resp,freq)
```

to create a SISO FRD model.

See Also

chgunits	Change units for an FRD model
frdata	Quick access to data for an FRD model
set	Set the properties for an LTI model
SS	Create state-space models

tf Create transfer functions
zpk Create zero-pole-gain models

Quick access to data for a frequency response data object

Syntax

```
[response,freq] = frdata(sys)
[response,freq,Ts] = frdata(sys)
[response,freq] = frdata(sys,'v')
```

Description

[response,freq] = frdata(sys) returns the response data and frequency samples of the FRD model sys. For an FRD model with Ny outputs and Nu inputs at Nf frequencies:

- response is an Ny-by-Nu-by-Nf multidimensional array where the (i,j) entry specifies the response from input j to output i.
- freq is a column vector of length Nf that contains the frequency samples of the FRD model.

See Table 11-14, "Data Format for the Argument response in FRD Models," on page 80 for more information on the data format for FRD response data.

For SISO FRD models, the syntax

```
[response, freq] = frdata(sys, 'v')
```

forces frdata to return the response data and frequencies directly as column vectors rather than as cell arrays (see example below).

[response, freq, Ts] = frdata(sys) also returns the sample time Ts.

Other properties of sys can be accessed with get or by direct structure-like referencing (e.g., sys.Units).

Arguments

The input argument sys to frdata must be an FRD model.

Example

Typing the commands

```
freq = logspace(1,2,2);
resp = .05*(freq).*exp(i*2*freq);
sys = frd(resp,freq);
[resp,freq] = frdata(sys,'v')
```

returns the FRD model data

```
resp = 0.2040 + 0.4565i
```

frdata

freqresp

Purpose

Compute frequency response over grid of frequencies

Syntax

H = freqresp(sys, w)

Description

H = freqresp(sys,w) computes the frequency response of the LTI model sys at the real frequency points specified by the vector w. The frequencies must be in radians/sec. For single LTI Models, freqresp(sys,w) returns a 3-D array H with the frequency as the last dimension (see "Arguments" below). For LTI arrays of size [Ny Nu S1 ... Sn], freqresp(sys,w) returns a [Ny-by-Nu-by-S1-by-...-by-Sn] length (w) array.

In continuous time, the response at a frequency ω is the transfer function value at $s = j\omega$. For state-space models, this value is given by

$$H(j\omega) = D + C(j\omega I - A)^{-1}B$$

In discrete time, the real frequencies $w(1), \dots, w(N)$ are mapped to points on the unit circle using the transformation $z = e^{j\omega T_s}$

where T_s is the sample time. The transfer function is then evaluated at the resulting z values. The default T_s = 1 is used for models with unspecified sample time.

Remark

If sys is an FRD model, freqresp(sys,w), w can only include frequencies in sys.frequency. Interpolation and extrapolation are not supported. To interpolate an FRD model, use interp.

Arguments

The output argument H is a 3-D array with dimensions

 $(number\ of\ outputs) \times (number\ of\ inputs) \times (length\ of\ w)$

For SISO systems, H(1,1,k) gives the scalar response at the frequency w(k). For MIMO systems, the frequency response at w(k) is H(:,:,k), a matrix with as many rows as outputs and as many columns as inputs.

Example

Compute the frequency response of

$$P(s) = \begin{bmatrix} 0 & \frac{1}{s+1} \\ \frac{s-1}{s+2} & 1 \end{bmatrix}$$

at the frequencies $\omega = 1, 10, 100$. Type

$$H(:,:,1) =$$

$$H(:,:,2) =$$

$$H(:,:,3) =$$

The three displayed matrices are the values of $P(j\omega)$ for

$$\omega = 1$$
, $\omega = 10$, $\omega = 100$

The third index in the 3-D array H is relative to the frequency vector w, so you can extract the frequency response at $\omega = 10$ rad/sec by

freqresp

Algorithm

For transfer functions or zero-pole-gain models, frequency evaluates the numerator(s) and denominator(s) at the specified frequency points. For continuous-time state-space models (A, B, C, D), the frequency response is

$$D + C(j\omega - A)^{-1}B$$
, $\omega = \omega_1, ..., \omega_N$

For efficiency, A is reduced to upper Hessenberg form and the linear equation $(j\omega - A)X = B$ is solved at each frequency point, taking advantage of the Hessenberg structure. The reduction to Hessenberg form provides a good compromise between efficiency and reliability. See [1] for more details on this technique.

Diagnostics

If the system has a pole on the $j\omega$ axis (or unit circle in the discrete-time case) and w happens to contain this frequency point, the gain is infinite, $j\omega I - A$ is singular, and freqresp produces the following warning message.

Singularity in freq. response due to jw-axis or unit circle pole.

References

[1] Laub, A.J., "Efficient Multivariable Frequency Response Computations," *IEEE Transactions on Automatic Control*, AC-26 (1981), pp. 407-408.

See Also

evalfr Response at single complex frequency bode Bode plot
nyquist Nyquist plot
nichols Nichols plot
sigma Singular value plot
ltiview LTI system viewer

interp Interpolate FRD model between frequency points

Generalized solver for continuous-time algebraic Riccati equations

Syntax

Description

[X,L,report] = gcare(H,J,ns) computes the unique stabilizing solution X of the continuous-time algebraic Riccati equation associated with a Hamiltonian pencil of the form

$$H - tJ = \begin{bmatrix} A & F & S1 \\ G & -A' & -S2 \\ S2' & S1' & R \end{bmatrix} - \begin{bmatrix} E & 0 & 0 \\ 0 & E' & 0 \\ 0 & 0 & 0 \end{bmatrix}$$

The optional input ns is the row size of the A matrix. Default values for J and ns correspond to E=I and R=[].

Optionally, gcare returns the vector L of closed-loop eigenvalues and a diagnosis report with value:

- −1 if the Hamiltonian pencil has *jw*-axis eigenvalues
- ullet –2 if there is no finite stabilizing solution X
- ullet 0 if a finite stabilizing solution X exists

This syntax does not issue any error message when X fails to exist.

[X1,X2,D,L] = gcare(H,..., factor) returns two matrices X1, X2 and a diagonal scaling matrix D such that X = D*(X2/X1)*D. The vector L contains the closed-loop eigenvalues. All outputs are empty when the associated Hamiltonian matrix has eigenvalues on the imaginary axis.

See Also

care gdare Solver for continuous-time algebraic Riccati equations Generalized solver for discrete-time algebraic Riccati equations

gdare

Purpose

Generalized solver for discrete-time algebraic Riccati equations.

Syntax

Description

[X,L,report] = gdare(H,J,ns) computes the unique stabilizing solution X of the discrete-time algebraic Riccati equation associated with a Symplectic pencil of the form

$$H - tJ = \begin{bmatrix} A & F & B \\ -Q & E' & -S \\ S' & 0 & R \end{bmatrix} - \begin{bmatrix} E & 0 & 0 \\ 0 & A' & 0 \\ 0 & -B' & 0 \end{bmatrix}$$

The third input ns is the row size of the *A* matrix.

Optionally, gdare returns the vector ${\tt L}$ of closed-loop eigenvalues and a diagnosis report with value:

- -1 if the Symplectic pencil has eigenvalues on the unit circle
- ullet -2 if there is no finite stabilizing solution X
- 0 if a finite stabilizing solution X exists

This syntax does not issue any error message when X fails to exist.

[X1,X2,D,L] = gdare(H,J,NS,'factor') returns two matrices X1, X2 and a diagonal scaling matrix D such that X = D*(X2/X1)*D. The vector L contains the closed-loop eigenvalues. All outputs are empty when the Symplectic pencil has eigenvalues on the unit circle.

See Also

dare Solver for discrete-time algebraic Riccati equations gcare Generalized solver for continuous-time algebraic Riccati equations

Generate test input signals for 1sim

Syntax

```
[u,t] = gensig(type,tau)
[u,t] = gensig(type,tau,Tf,Ts)
```

Description

[u,t] = gensig(type,tau) generates a scalar signal u of class type and with period tau (in seconds). The following types of signals are available.

'sin' Sine wave.
'square' Square wave.
'pulse' Periodic pulse.

gensig returns a vector t of time samples and the vector u of signal values at these samples. All generated signals have unit amplitude.

[u,t] = gensig(type,tau,Tf,Ts) also specifies the time duration Tf of the signal and the spacing Ts between the time samples t.

You can feed the outputs u and t directly to 1sim and simulate the response of a single-input linear system to the specified signal. Since t is uniquely determined by Tf and Ts, you can also generate inputs for multi-input systems by repeated calls to gensig.

Example

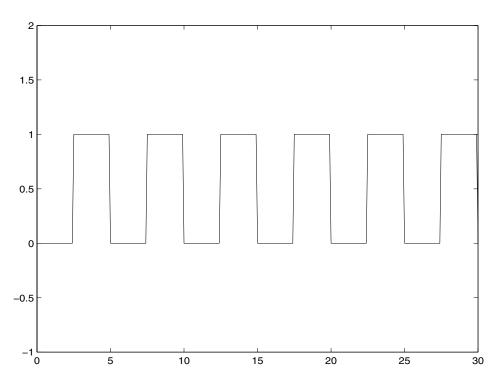
Generate a square wave with period 5 seconds, duration 30 seconds, and sampling every 0.1 second.

```
[u,t] = gensig('square',5,30,0.1)
```

Plot the resulting signal.

plot(t,u)





See Also

lsim

Simulate response to arbitrary inputs

Access/query LTI property values

Syntax

```
Value = get(sys,'PropertyName')
get(sys)
Struct = get(sys)
```

Description

Value = get(sys, 'PropertyName') returns the current value of the property PropertyName of the LTI model sys. The string 'PropertyName' can be the full property name (for example, 'UserData') or any unambiguous case-insensitive abbreviation (for example, 'user'). You can specify any generic LTI property, or any property specific to the model sys (see "LTI Properties" for details on generic and model-specific LTI properties).

Struct = get(sys) converts the TF, SS, or ZPK object sys into a standard MATLAB structure with the property names as field names and the property values as field values.

Without left-side argument,

```
get(sys)
```

displays all properties of sys and their values.

Example

Consider the discrete-time SISO transfer function defined by

```
h = tf(1,[1 2],0.1,'inputname','voltage','user','hello')
```

You can display all LTI properties of h with

```
get(h)
    num = {[0 1]}
    den = {[1 2]}
    Variable = 'z'
    Ts = 0.1
    InputDelay = 0
    OutputDelay = 0
    ioDelay = 0
    InputName = {'voltage'}
    OutputName = {''}
    InputGroup = {0x2 cell}
    OutputGroup = {0x2 cell}
```

```
Notes = {}
           UserData = 'hello'
or query only about the numerator and sample time values by
  get(h,'num')
  ans =
       [1x2 double]
and
  get(h,'ts')
  ans =
       0.1000
```

Because the numerator data (num property) is always stored as a cell array, the first command evaluates to a cell array containing the row vector [0 1].

Remark

An alternative to the syntax

```
Value = get(sys,'PropertyName')
```

is the structure-like referencing

```
Value = sys.PropertyName
```

For example,

sys.Ts sys.a

sys.user

return the values of the sample time, A matrix, and UserData property of the (state-space) model sys.

See Also

Quick access to frequency response data
Set/modify LTI properties
Quick access to state-space data
Quick access to transfer function data
Quick access to zero-pole-gain data

Purpose Return @PlotOptions handles or plot options property

Syntax p = getoptions(h)

p = getoptions(h,propertyname)

Description p = getoptions(h) returns the plot options handle associated with plot handle

h. p contains all the settable options for a given response plot.

p = getoptions(h,propertyname) returns the specified options property, propertyname, for the plot with handle h. You can use this to interrogate a plot

handle. For example,

p = getoptions(h, 'Grid')

returns 'on' if a grid is visible, and 'off' when it is not.

See Also setoptions Set plot options for response plots

Compute controllability and observability grammians

Syntax

Description

gram calculates controllability and observability grammians. You can use grammians to study the controllability and observability properties of state-space models and for model reduction [1,2]. They have better numerical properties than the controllability and observability matrices formed by ctrb and obsv.

Given the continuous-time state-space model

$$\dot{x} = Ax + Bu$$

$$v = Cx + Du$$

the controllability grammian is defined by

$$W_c = \int_0^\infty e^{A\tau} B B^T e^{A^T \tau} d\tau$$

and the observability grammian by

$$W_o = \int_0^\infty e^{A^T \tau} C^T C e^{A \tau} d\tau$$

The discrete-time counterparts are

$$W_c = \sum_{k=0}^{\infty} A^k B B^T (A^T)^k, \qquad W_o = \sum_{k=0}^{\infty} (A^T)^k C^T C A^k$$

The controllability grammian is positive definite if and only if (A, B) is controllable. Similarly, the observability grammian is positive definite if and only if (C, A) is observable.

Use the commands

```
Wc = gram(sys,'c') % controllability grammian
Wo = gram(sys,'o') % observability grammian
```

to compute the grammians of a continuous or discrete system. The LTI model sys must be in state-space form.

Algorithm

The controllability grammian \boldsymbol{W}_c is obtained by solving the continuous-time Lyapunov equation

$$AW_c + W_c A^T + BB^T = 0$$

or its discrete-time counterpart

$$AW_cA^T - W_c + BB^T = 0$$

Similarly, the observability grammian W_o solves the Lyapunov equation

$$A^T W_o + W_o A + C^T C = 0$$

in continuous time, and the Lyapunov equation

$$A^T W_o A - W_o + C^T C = 0$$

in discrete time.

Limitations

The A matrix must be stable (all eigenvalues have negative real part in continuous time, and magnitude strictly less than one in discrete time).

References

[1] Kailath, T., Linear Systems, Prentice-Hall, 1980.

See Also

balreal Grammian-based balancing of state-space realizations
ctrb Controllability matrix
lyap, dlyap Lyapunov equation solvers
obsv Observability matrix

hasdelay

Purpose Test if an LTI model has time delays

Syntax hasdelay(sys)

Description hasdelay (sys) returns 1 (true) if the LTI model sys has input delays, output

delays, or I/O delays, and 0 (false) otherwise.

See Also delay2z Changes transfer functions of discrete-time LTI models

with delays to rational functions or absorbs FRD delays

into the frequency response phase information

totaldelay Combines delays for an LTI model

Purpose

Computes the Hankel singular values of an LTI model

Syntax

```
hsv = hsvd(sys)
hsvd(sys)
[hsv,baldata] = hsvd(sys)
```

Description

hsv = hsvd(sys) computes the Hankel singular values hsv of the LTI model sys. In state coordinates that equalize the input-to-state and state-to-output energy transfers, the Hankel singular values measure the contribution of each state to the input/output behavior. Hankel singular values are to model order what singular values are to matrix rank. In particular, small Hankel singular values signal states that can be discarded to simplify the model (see balred).

For models with unstable poles, hsvd only computes the Hankel singular values of the stable part and entries of hsv corresponding to unstable modes are set to Inf. Use

to specify additional options for the stable/unstable decomposition, see STABSEP for details. The default values are ATOL=0, RTOL=1e-8, and ALPHA=1e-8.

hsvd(sys) displays a plot of the Hankel singular values.

[hsv,baldata] = hsvd(sys) returns additional data to speed up model order reduction with balred. For example

computes 3 approximations of sys of orders 8, 9, 10.

There is more than one hsvd available. Type

```
help lti/hsvd
```

for more information.

Algorithm

The AbsTo1, RelTo1, and ALPHA parameters are only used for models with unstable or marginally stable dynamics. Because Hankel singular values are

only meaningful for stable dynamics, hsvd must first splitsuch models into the sum of their stable and unstable parts:

$$G = G_s + G_ns$$

This decomposition can be tricky when the model has modes close to the stability boundary (e.g., a pole at s=-1e-10), or clusters of modes on the stability boundary (e.g., double or triple integrators). While hsvd is able to overcome these difficulties in most cases, it sometimes produces unexpected results such as

- 1 Large Hankel singular values for the stable part.
 This happens when the stable part G_s contains some poles very close to the stability boundary. To force such modes into the unstable group, increase the 'Offset' option to slightly grow the unstable region.
- **2** Too many modes are labeled "unstable." For example, you see 5 red bars in the HSV plot when your model had only 2 unstable poles.

The stable/unstable decomposition algorithm has built-in accuracy checks that reject decompositions causing a significant loss of accuracy in the frequency response. Such loss of accuracy arises, e.g., when trying to split a cluster of stable and unstable modes near s=0. Because such clusters are numerically equivalent to a multiple pole at s=0, it is actually desirable to treat the whole cluster as unstable. In some cases, however, large relative errors in low-gain frequency bands can trip the accuracy checks and lead to a rejection of valid decompositions. Additional modes are then absorbed into the unstable part G_ns, unduly increasing its order.

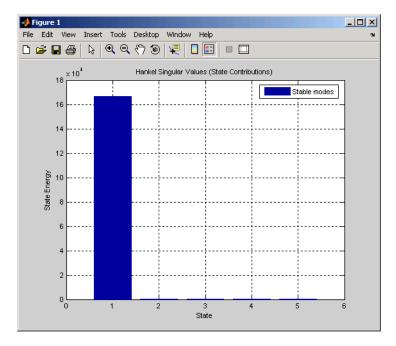
Such issues can be easily corrected by adjusting the AbsTol and RelTol tolerances. By setting AbsTol to a fraction of smallest gain of interest in your model, you tell the algorithm to ignore errors below a certain gain threshold. By increasing RelTol, you tell the algorithm to sacrifice some relative model accuracy in exchange for keeping more modes in the stable part G s.

Examples

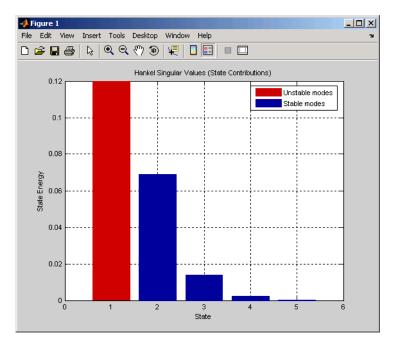
These examples illustrate the use of AbsTol and offset.

Example 1: Large Hankel singular values for the stable part.

First, create a system with a stable pole very near to 0, then calculate the Hankel singular values.



Notice the dominant Hankel singular value with 1e5 magnitude, due to the mode s=-1e-7 near the imaginary axis. Set the offset=1e-6 to treat this mode as unstable



The dominant Hankel singular value is now shown as unstable.

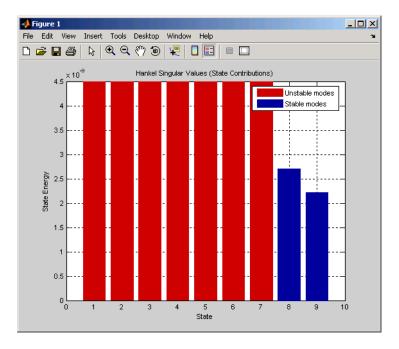
Example 2: Too many modes are labeled as unstable.

Create a system with three unstable modes. Then calculate the Hankel singular values. $\,$

- -1.0000e+000 -1.0000e+001i
- -2.0000e+000

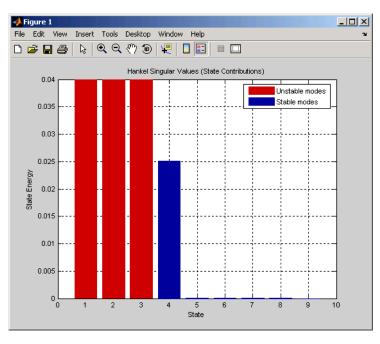
There are 3 unstable modes, but there are 7 "unstable" Hankel singular values on the plot.

hsvd(sys)



Note the low gain of -400 dB near to pole s=1e3 (w=1e3 rad/s). Try increasing the absolute tolerance to AbsTol = 1e-16 (=-320 dB).

hsvd(sys,'AbsTol',1e-16)



This fixed the problem, as this figure shows.

There are now only three unstable modes, the corrrect number for the system sys.

See Also

balred balreal Model order reduction Gramian-based balancing of state-space realizations

Purpose

Plot the Hankel singular values and return the plot handle

Syntax

```
h = hsvplot(sys, 'AbsTol',ATOL,'RelTol',RTOL,'Offset',ALPHA)
h = hsvplot(AX,sys,...)
```

Description

h = hsvplot(sys) plots the Hankel singular values of an LTI system sys and returns the plot handle h. You can use this handle to customize the plot with the getoptions and setoptions commands. Type

```
help hsvoptions
```

h = hsvplot(sys);

for a list of available plot options.

hsvplot(sys) plots the Hankel singular values of the LTI model sys. See hsvd for details on the meaning and purpose of Hankel singular values. The Hankel singular values for the stable and unstable modes of sys are shown in blue and red, respectively.

hsvplot(sys, AbsTol',ATOL,'RelTol',RTOL,'Offset',ALPHA) specifies additional options for computing the Hankel singular values.

hsvplot(AX, sys,...) attaches the plot to the axes with handle AX.

Example

Use the plot handle to change plot options in the Hankel singular values plot.

```
sys = rss(20);
h = hsvplot(sys, 'AbsTol',1e-6);
% Switch to log scale and modify Offset parameter
setoptions(h, 'Yscale', 'log', 'Offset',0.3)
```

See Also

getoptions Get plot options

hsvd Plot Hankel singular values (does not return the

handle)

setoptions Set plot options

Purpose

Compute the impulse response of LTI models

Syntax

```
impulse(sys)
impulse(sys,t)

impulse(sys1,sys2,...,sysN)
impulse(sys1,sys2,...,sysN,t)
impulse(sys1,'PlotStyle1',...,sysN,'PlotStyleN')

[y,t,x] = impulse(sys)
```

Description

impulse calculates the unit impulse response of a linear system. The impulse response is the response to a Dirac input $\delta(t)$ for continuous-time systems and to a unit pulse at t=0 for discrete-time systems. Zero initial state is assumed in the state-space case. When invoked without left-hand arguments, this function plots the impulse response on the screen.

impulse(sys) plots the impulse response of an arbitrary LTI model sys. This model can be continuous or discrete, and SISO or MIMO. The impulse response of multi-input systems is the collection of impulse responses for each input channel. The duration of simulation is determined automatically to display the transient behavior of the response.

impulse(sys,t) sets the simulation horizon explicitly. You can specify either
a final time t = Tfinal (in seconds), or a vector of evenly spaced time samples
of the form

```
t = 0:dt:Tfinal
```

For discrete systems, the spacing dt should match the sample period. For continuous systems, dt becomes the sample time of the discretized simulation model (see "Algorithm"), so make sure to choose dt small enough to capture transient phenomena.

To plot the impulse responses of several LTI models sys1,..., sysN on a single figure, use

```
impulse(sys1,sys2,...,sysN)
impulse(sys1,sys2,...,sysN,t)
```

As with bode or plot, you can specify a particular color, linestyle, and/or marker for each system, for example,

```
impulse(sys1, 'y:', sys2, 'g--')
```

See "Plotting and Comparing Multiple Systems" and the bode entry in this section for more details.

When invoked with left-side arguments,

```
[y,t] = impulse(sys)
[y,t,x] = impulse(sys) % for state-space models only
y = impulse(sys,t)
```

return the output response y, the time vector t used for simulation, and the state trajectories x (for state-space models only). No plot is drawn on the screen. For single-input systems, y has as many rows as time samples (length of t), and as many columns as outputs. In the multi-input case, the impulse responses of each input channel are stacked up along the third dimension of y. The dimensions of y are then

```
(length of t) \times (number of outputs) \times (number of inputs)
```

and y(:,:,j) gives the response to an impulse disturbance entering the jth input channel. Similarly, the dimensions of x are

(length of t)
$$\times$$
 (number of states) \times (number of inputs)

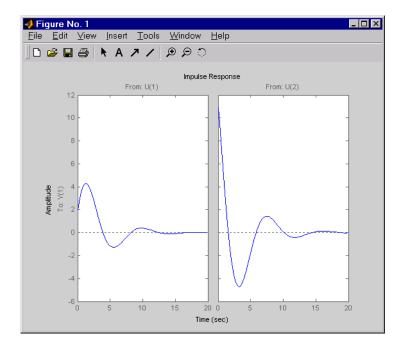
Example

To plot the impulse response of the second-order state-space model

$$\begin{bmatrix} \dot{x}_1 \\ \dot{x}_2 \end{bmatrix} = \begin{bmatrix} -0.5572 & -0.7814 \\ 0.7814 & 0 \end{bmatrix} \begin{bmatrix} x_1 \\ x_2 \end{bmatrix} + \begin{bmatrix} 1 & -1 \\ 0 & 2 \end{bmatrix} \begin{bmatrix} u_1 \\ u_2 \end{bmatrix}$$
$$y = \begin{bmatrix} 1.9691 & 6.4493 \end{bmatrix} \begin{bmatrix} x_1 \\ x_2 \end{bmatrix}$$

use the following commands.

```
sys = ss(a,b,c,0);
impulse(sys)
```



The left plot shows the impulse response of the first input channel, and the right plot shows the impulse response of the second input channel.

You can store the impulse response data in MATLAB arrays by

$$[y,t] = impulse(sys)$$

Because this system has two inputs, y is a 3-D array with dimensions

(the first dimension is the length of t). The impulse response of the first input channel is then accessed by

```
y(:,:,1)
```

Algorithm

Continuous-time models are first converted to state space. The impulse response of a single-input state-space model

$$\dot{x} = Ax + bu$$

$$y = Cx$$

is equivalent to the following unforced response with initial state $\,b\,$.

$$\dot{x} = Ax, \qquad x(0) = b
y = Cx$$

To simulate this response, the system is discretized using zero-order hold on the inputs. The sampling period is chosen automatically based on the system dynamics, except when a time vector t = 0:dt:Tf is supplied (dt is then used as sampling period).

Limitations

The impulse response of a continuous system with nonzero D matrix is infinite at t=0. impulse ignores this discontinuity and returns the lower continuity value Cb at t=0.

See Also

ltiview LTI system viewer step Step response

initial Free response to initial condition
1sim Simulate response to arbitrary inputs

impulseplot

Purpose

Compute the impulse response and return the plot handle

Syntax

```
h = impulseplot(sys,Tfinal)
h = impulseplot(sys,t)
h = impulseplot(sys1,sys2,...,t)
h = impulseplot(AX,...)
h = impulseplot(..., plotoptions)
```

Description

h = impulseplot(sys) plots the impulse response of the LTI model sys (created with either tf, zpk, or ss). For multiinput models, independent impulse commands are applied to each input channel. The time range and number of points are chosen automatically. For continuous systems with direct feedthrough, the infinite pulse at t=0 is disregarded. impulseplot also returns the plot handle, h. You can use this handle to customize the plot with the getoptions and setoptions commands. Type

```
help timeoptions
```

h = impulseplot(sys)

for a list of available plot options.

impulseplot(sys) plots the impulse response of the LTI model without returning the plot handle.

impulseplot(sys, Tfinal) simulates the impulse response from t=0 to the final time t=Tfinal. For discrete-time systems with unspecified sampling time, Tfinal is interpreted as the number of samples.

impulseplot(sys,t) uses the user-supplied time vector t for simulation. For discrete-time models, t should be of the form Ti:Ts:Tf, where Ts is the sample time. For continuous-time models, t should be of the form Ti:dt:Tf, where dt becomes the sample time of a discrete approximation to the continuous system. The impulse is always assumed to arise at t=0 (regardless of Ti).

impulseplot(sys1,sys2,...,t) plots the impulse response of multiple LTI models sys1,sys2,... on a single plot. The time vector t is optional. You can also specify a color, line style, and marker for each system, as in

```
impulseplot(sys1,'r',sys2,'y--',sys3,'gx')
```

impulseplot

impulseplot (AX,...) plots into the axes with handle AX.

impulseplot(..., plotoptions) plots the impulse response with the options specified in plotoptions. Type

```
help timeoptions
```

for more detail.

Example

Normalize the impulse response of a third-order system.

```
sys = rss(3);
h = impulseplot(sys);
% Normalize responses
setoptions(h,'Normalize','on');
```

See Also

getoptions Get plot options

impulse Plot impulse responses

setoptions Set plot options

Purpose

Compute the initial condition response of state-space models

Syntax

```
initial(sys,x0)
initial(sys,x0,t)

initial(sys1,sys2,...,sysN,x0)
initial(sys1,sys2,...,sysN,x0,t)
initial(sys1,'PlotStyle1',...,sysN,'PlotStyleN',x0)

[y,t,x] = initial(sys,x0)
```

Description

initial calculates the unforced response of a state-space model with an initial condition on the states.

$$\dot{x} = Ax , \qquad x(0) = x_0$$

$$v = Cx$$

This function is applicable to either continuous- or discrete-time models. When invoked without left-side arguments, initial plots the initial condition response on the screen.

initial(sys,x0) plots the response of sys to an initial condition x0 on the states. sys can be any *state-space* model (continuous or discrete, SISO or MIMO, with or without inputs). The duration of simulation is determined automatically to reflect adequately the response transients.

initial(sys,x0,t) explicitly sets the simulation horizon. You can specify
either a final time t = Tfinal (in seconds), or a vector of evenly spaced time
samples of the form

```
t = 0:dt:Tfinal
```

For discrete systems, the spacing dt should match the sample period. For continuous systems, dt becomes the sample time of the discretized simulation model (see impulse), so make sure to choose dt small enough to capture transient phenomena.

To plot the initial condition responses of several LTI models on a single figure, use

```
initial(sys1,sys2,...,sysN,x0)
initial(sys1,sys2,...,sysN,x0,t)
```

(see impulse for details).

When invoked with left-side arguments,

return the output response y, the time vector t used for simulation, and the state trajectories x. No plot is drawn on the screen. The array y has as many rows as time samples (length of t) and as many columns as outputs. Similarly, x has length(t) rows and as many columns as states.

Example

Plot the response of the state-space model

$$\begin{bmatrix} \dot{x}_1 \\ \dot{x}_2 \end{bmatrix} = \begin{bmatrix} -0.5572 & -0.7814 \\ 0.7814 & 0 \end{bmatrix} \begin{bmatrix} x_1 \\ x_2 \end{bmatrix}$$
$$y = \begin{bmatrix} 1.9691 & 6.4493 \end{bmatrix} \begin{bmatrix} x_1 \\ x_2 \end{bmatrix}$$

to the initial condition

$$x(0) = \begin{bmatrix} 1 \\ 0 \end{bmatrix}$$

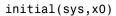
$$a = \begin{bmatrix} -0.5572 & -0.7814; 0.7814 & 0 \end{bmatrix};$$

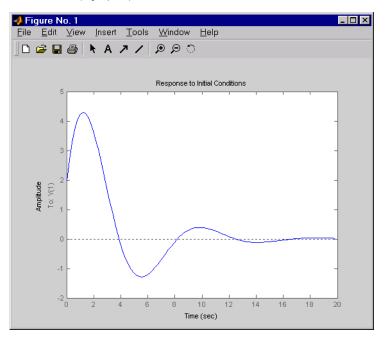
$$c = \begin{bmatrix} 1.9691 & 6.4493 \end{bmatrix};$$

$$x0 = \begin{bmatrix} 1 & ; & 0 \end{bmatrix}$$

$$sys = ss(a, [], c, []);$$

initial





See Also

impulse	Impulse response
lsim	Simulate response to arbitrary inputs
ltiview	LTI system viewer
step	Step response

Purpose

Compute initial condition responses and return plot handles

Syntax

```
h = initialplot(sys, x0)
h = initialplot(sys,x0,Tfinal)
h = initialplot(sys,x0,t)
h = initialplot(sys1,sys2,...,x0,t)
h = initialplot(AX,...)
h = initialplot(..., plotoptions)
```

Description

initialplot(sys,x0) plots the undriven response of the state-space model sys (created with ss) with initial condition x0 on the states. This response is characterized by these equations:

```
Continuous time: x = A x, y = C x, x(0) = x0
Discrete time: x[k+1] = A x[k], y[k] = C x[k], x[0] = x0
```

The time range and number of points are chosen automatically. initialplot also returns the plot handle h. You can use this handle to customize the plot with the getoptions and setoptions commands. Type

```
help timeoptions
```

for a list of available plot options.

initialplot(sys,x0,Tfinal) simulates the time response from t=0 to the final time t=Tfinal. For discrete-time models with unspecified sample time, Tfinal should be the number of samples.

initialplot(sys,x0,t) specifies a time vector t to be used for simulation. For discrete systems, t should be of the form 0:Ts:Tf, where Ts is the sample time. For continuous-time models, t should be of the form 0:dt:Tf, where dt becomes the sample time of a discrete approximation of the continuous model.

initialplot(sys1,sys2,...,x0,t) plots the response of multiple LTI models sys1,sys2,... on a single plot. The time vector t is optional. You can also specify a color, line style, and marker for each system, as in

```
initialplot(sys1,'r',sys2,'y--',sys3,'gx',x0).
```

initialplot(AX,...) plots into the axes with handle AX.

initialplot

initialplot(..., plotoptions) plots the initial condition response with the options specified in plotoptions. Type

```
help timeoptions
```

for more detail.

Example

Plot a third-order system's response to initial conditions and use the plot handle to change the plot's title.

```
sys = rss(3);
h = initialplot(sys,[1,1,1])
p = getoptions(h); % Get options for plot.
p.Title.String = 'My Title'; % Change title in options.
setoptions(h,p); % Apply options to the plot.
```

See Also

getoptions Get plot options

initial Plot response to initial conditions

setoptions Set plot options

Purpose Interpolate an FRD model between frequency points

Syntax isys = interp(sys, freqs) interpolates the frequency response data

contained in the FRD model sys at the frequencies freqs. interp, which is an overloaded version of the MATLAB function interp, uses linear interpolation and returns an FRD model isys containing the interpolated data at the new

frequencies freqs.

You should express the frequency values freqs in the same units as sys.frequency. The frequency values must lie between the smallest and

largest frequency points in sys (extrapolation is not supported).

See Also frequency response of LTI models

ltimodels Help on LTI models

inv

Purpose

Invert LTI systems

Syntax

isys = inv(sys)

Description

inv inverts the input/output relation

$$y = G(s)u$$

to produce the LTI system with the transfer matrix $H(s) = G(s)^{-1}$.

$$u = H(s)y$$

This operation is defined only for square systems (same number of inputs and outputs) with an invertible feedthrough matrix D. inv handles both continuous- and discrete-time systems.

Example

Consider

$$H(s) = \begin{bmatrix} 1 & \frac{1}{s+1} \\ 0 & 1 \end{bmatrix}$$

At the MATLAB prompt, type

$$H = [1 tf(1,[1 1]);0 1]$$

 $Hi = inv(H)$

to invert it. MATLAB returns

Transfer function from input 1 to output...

#1: 1

#2: 0

Transfer function from input 2 to output...

#1: ----s + 1

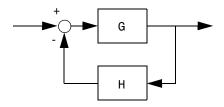
#2: 1

You can verify that

is the identity transfer function (static gain I).

Limitations

Do not use inv to model feedback connections such as



While it seems reasonable to evaluate the corresponding closed-loop transfer function $\left(I+GH\right)^{-1}G$ as

$$inv(1+g*h) * g$$

this typically leads to nonminimal closed-loop models. For example,

yields a third-order closed-loop model with an unstable pole-zero cancellation at s=1.

cloop

Use feedback to avoid such pitfalls.

cloop = feedback(g,h)

Zero/pole/gain:

(s^2 + s + 1)

iopzmap

Purpose Plot pole-zero maps for I/O pairs of LTI models

Syntax iopzmap(sys)

iopzmap(sys1,sys2,...)

Description

iopzmap(sys) computes and plots the poles and zeros of each input/output pair of the LTI model sys. The poles are plotted as x's and the zeros are plotted as o's.

iopzmap(sys1, sys2,...) shows the poles and zeros of multiple LTI models sys1,sys2,... on a single plot. You can specify distinctive colors for each model, as in iopzmap(sys1, 'r', sys2, 'y', sys3, 'g').

The functions sgrid or zgrid can be used to plot lines of constant damping ratio and natural frequency in the *s* or *z* plane.

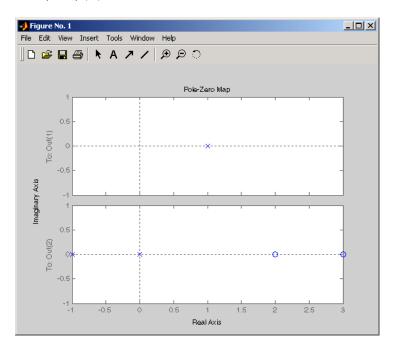
For arrays sys of LTI models, iopzmap plots the poles and zeros o each model in the array on the same diagram.

Example

Create a one-input, two-output system and plot pole-zero maps for I/O pairs.

H = [tf(-5, [1 -1]); tf([1 -5 6], [1 1 0])];

iopzmap(H)



See Also

pzmap	Pole-zero map
pole	Compute system poles
zero	Compute system zeros
sgrid	Grid for s-plane plots
zgrid	Grid for z -plane plots
ltimodels	Information about LTI models

Purpose

Plot pole-zero maps for I/O pairs and return the plot handle

Syntax

```
h = iopzplot(sys1,sys2,...)
h = iopzplot(AX,...)
h = iopzplot(..., plotoptions)
```

Description

h = iopzplot(sys) computes and plots the poles and zeros of each input/ output pair of the LTI model SYS. The poles are plotted as x's and the zeros are plotted as o's. It also returns the plot handle h. You can use this handle to customize the plot with the getoptions and setoptions commands. Type

```
help pzoptions
```

h = iopzplot(svs)

for a list of available plot options.

iopzplot(sys1, sys2,...) shows the poles and zeros of multiple LTI models SYS1,SYS2,... on a single plot. You can specify distinctive colors for each model, as in

```
iopzplot(sys1,'r',sys2,'y',sys3,'g')
```

iopzplot(AX,...) plots into the axes with handle AX.

iopzplot(..., plotoptions) plots the poles and zeros with the options specified in plotoptions. Type

```
help pzoptions
```

for more detail.

The function sgrid or zgrid can be used to plot lines of constant damping ratio and natural frequency in the s or z plane.

For arrays sys of LTI models, iopzplot plots the poles and zeros of each model in the array on the same diagram.

Example

Use the plot handle to change the I/O grouping of a pole/zero map.

```
sys = rss(3,2,2);
h = iopzplot(sys);
% View all input-output pairs on a single axis.
```

iopzplot

setoptions(h,'IOGrouping','all')

See Also

getoptions Get plot options

iopzmap Plot a pole/zero map of I/O pairs

setoptions Set plot options

isct, isdt

Purpose

Determine whether an LTI model is continuous or discrete

Syntax

```
boo = isct(sys)
boo = isdt(sys)
```

Description

boo = isct(sys) returns 1 (true) if the LTI model sys is continuous and 0 (false) otherwise. sys is continuous if its sample time is zero, that is, sys.Ts=0.

boo = isdt(sys) returns 1 (true) if sys is discrete and 0 (false) otherwise. Discrete-time LTI models have a nonzero sample time, except for empty models and static gains, which are regarded as either continuous or discrete as long as their sample time is not explicitly set to a nonzero value. Thus both

```
isct(tf(10))
isdt(tf(10))
```

are true. However, if you explicitly label a gain as discrete, for example, by typing

```
g = tf(10, 'ts', 0.01)
```

isct(g) now returns false and only isdt(g) is true.

See Also

isa	Determine LTI model type
isempty	True for empty LTI models
isproper	True for proper LTI models

Purpose Test if an LTI model is empty

Syntax boo = isempty(sys)

Description isempty (sys) returns 1 (true) if the LTI model sys has no input or no output,

and 0 (false) otherwise.

Example Both commands

isempty(tf) % tf by itself returns an empty transfer function

isempty(ss(1,2,[],[]))

return 1 (true) while

isempty(ss(1,2,3,4))

returns 0 (false).

See Also issiso True for SISO systems

size I/O dimensions and array dimensions of LTI models

isproper

Purpose Test if an LTI model is proper

Syntax boo = isproper(sys)

Description isproper (sys) returns 1 (true) if the LTI model sys is proper and 0 (false)

otherwise.

State-space models are always proper. SISO transfer functions or $\,$

zero-pole-gain models are proper if the degree of their numerator is less than or equal to the degree of their denominator. MIMO transfer functions are

proper if all their SISO entries are proper.

Example The following commands

return false and true, respectively.

issiso

Purpose Test if an LTI model is single-input/single-output (SISO)

Syntax boo = issiso(sys)

Description issiso(sys) returns 1 (true) if the LTI model sys is SISO and 0 (false)

otherwise.

See Also isempty True for empty LTI models

size I/O dimensions and array dimensions of LTI models

kalman

Purpose

Design continuous- or discrete-time Kalman estimator

Syntax

Description

kalman designs a Kalman state estimator given a state-space model of the plant and the process and measurement noise covariance data. The Kalman estimator is the optimal solution to the following continuous or discrete estimation problems.

Continuous-Time Estimation

Given the continuous plant

$$\dot{x} = Ax + Bu + Gw$$
 (state equation)
 $y_n = Cx + Du + Hw + v$ (measurement equation)

with known inputs u and process and measurement white noise w, v satisfying

$$E(w) = E(v) = 0$$
, $E(ww^{T}) = Q$, $E(vv^{T}) = R$, $E(wv^{T}) = N$

construct a state estimate $\hat{x}(t)$ that minimizes the steady-state error covariance

$$P = \lim_{t \to \infty} E(\{x - \hat{x}\}\{x - \hat{x}\}^T)$$

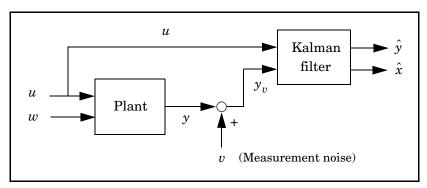
The optimal solution is the Kalman filter with equations

$$\hat{x} = A\hat{x} + Bu + L(y_v - C\hat{x} - Du)$$

$$\begin{bmatrix} \hat{y} \\ \hat{x} \end{bmatrix} = \begin{bmatrix} C \\ I \end{bmatrix} \hat{x} + \begin{bmatrix} D \\ 0 \end{bmatrix} u$$

where the filter gain L is determined by solving an algebraic Riccati equation. This estimator uses the known inputs u and the measurements y_n to generate the output and state estimates $y\,$ and $x\,$. Note that $y\,$ estimates the true plant output

$$y = Cx + Du + Hw$$



Kalman estimator

Discrete-Time Estimation

Given the discrete plant

$$x[n+1] = Ax[n] + Bu[n] + Gw[n]$$

$$y_v[n] = Cx[n] + Du[n] + Hw[n] + v[n]$$

and the noise covariance data

$$E(w[n]w[n]^T) = Q$$
, $E(v[n]v[n]^T) = R$, $E(w[n]v[n]^T) = N$

the Kalman estimator has equations

$$\begin{split} \hat{x}[n+1|n] &= A\hat{x}[n|n-1] + Bu[n] + L(y_v[n] - C\hat{x}[n|n-1] - Du[n]) \\ \begin{bmatrix} \hat{y}[n|n] \\ \hat{x}[n|n] \end{bmatrix} &= \begin{bmatrix} C(I-MC) \\ I-MC \end{bmatrix} \hat{x}[n|n-1] + \begin{bmatrix} (I-CM)D & CM \\ -MD & M \end{bmatrix} \begin{bmatrix} u[n] \\ y_v[n] \end{bmatrix} \end{split}$$

and generates optimal "current" output and state estimates y[n|n] and x[n|n] using all available measurements including $y_v[n]$. The gain matrices L and M are derived by solving a discrete Riccati equation. The *innovation gain* M is used to update the prediction $\hat{x}[n|n-1]$ using the new measurement $y_v[n]$.

$$\hat{x}[n|n] = \hat{x}[n|n-1] + M(\underbrace{y_v[n] - C\hat{x}[n|n-1] - Du[n]}_{\text{innovation}})$$

Usage

[kest,L,P] = kalman(sys,Qn,Rn,Nn) returns a state-space model kest of the Kalman estimator given the plant model sys and the noise covariance data Qn, Rn, Nn (matrices Q, R, N above). sys must be a state-space model with matrices

The resulting estimator kest has $[u \; ; y_v]$ as inputs and $[\hat{y} \; ; \hat{x}]$ (or their discrete-time counterparts) as outputs. You can omit the last input argument Nn when N=0.

The function kalman handles both continuous and discrete problems and produces a continuous estimator when sys is continuous, and a discrete estimator otherwise. In continuous time, kalman also returns the Kalman gain L and the steady-state error covariance matrix P. Note that P is the solution of the associated Riccati equation. In discrete time, the syntax

$$[kest,L,P,M,Z] = kalman(sys,Qn,Rn,Nn)$$

returns the filter gain ${\cal L}$ and innovations gain ${\cal M}$, as well as the steady-state error covariances

$$P = \lim_{n \to \infty} E(e[n|n-1]e[n|n-1]^T) , \qquad e[n|n-1] = x[n] - x[n|n-1]$$

$$Z = \lim_{n \to \infty} E(e[n|n]e[n|n]^T) , \qquad e[n|n] = x[n] - x[n|n]$$

Finally, use the syntaxes

for more general plants sys where the known inputs u and stochastic inputs w are mixed together, and not all outputs are measured. The index vectors sensors and known then specify which outputs y of sys are measured and which inputs u are known. All other inputs are assumed stochastic.

Example

See "LQG Design for the x-Axis" and "Kalman Filtering" for examples that use the kalman function.

Limitations

The plant and noise data must satisfy:

- (C, A) detectable
- $\overline{R} > 0$ and $\overline{Q} \overline{N} \overline{R}^{-1} \overline{N}^T \ge 0$
- $(A \overline{N}\overline{R}^{-1}C, \overline{Q} \overline{N}\overline{R}^{-1}\overline{N}^T)$ has no uncontrollable mode on the imaginary axis (or unit circle in discrete time)

with the notation

$$\overline{Q} = GQG^{T}$$

$$\overline{R} = R + HN + N^{T}H^{T} + HQH^{T}$$

$$\overline{N} = G(QH^{T} + N)$$

References

[1] Franklin, G.F., J.D. Powell, and M.L. Workman, *Digital Control of Dynamic Systems*, Second Edition, Addison-Wesley, 1990.

See Also

care	Solve continuous-time Riccati equations
dare	Solve discrete-time Riccati equations
estim	Form estimator given estimator gain
kalmd	Discrete Kalman estimator for continuous plant
lqgreg	Assemble LQG regulator
lqr	Design state-feedback LQ regulator

kalmd

Purpose

Design discrete Kalman estimator for continuous plant

Syntax

$$[kest,L,P,M,Z] = kalmd(sys,Qn,Rn,Ts)$$

Description

kalmd designs a discrete-time Kalman estimator that has response characteristics similar to a continuous-time estimator designed with kalman. This command is useful to derive a discrete estimator for digital implementation after a satisfactory continuous estimator has been designed.

[kest,L,P,M,Z] = kalmd(sys,Qn,Rn,Ts) produces a discrete Kalman
estimator kest with sample time Ts for the continuous-time plant

$$\dot{x} = Ax + Bu + Gw$$
 (state equation)
 $y_v = Cx + Du + v$ (measurement equation)

with process noise w and measurement noise v satisfying

$$E(w) = E(v) = 0$$
, $E(ww^{T}) = Q_n$, $E(vv^{T}) = R_n$, $E(wv^{T}) = 0$

The estimator kest is derived as follows. The continuous plant sys is first discretized using zero-order hold with sample time Ts (see c2d entry), and the continuous noise covariance matrices \boldsymbol{Q}_n and \boldsymbol{R}_n are replaced by their discrete equivalents

$$\begin{aligned} Q_d &= \int_0^{T_s} e^{A\tau} G Q G^T e^{A^T \tau} d\tau \\ R_d &= R/T_s \end{aligned}$$

The integral is computed using the matrix exponential formulas in [2]. A discrete-time estimator is then designed for the discretized plant and noise. See kalman for details on discrete-time Kalman estimation.

kalmd also returns the estimator gains L and M, and the discrete error covariance matrices P and Z (see kalman for details).

Limitations

The discretized problem data should satisfy the requirements for kalman.

References [1] Franklin, G.F., J.D. Powell, and M.L. Workman, *Digital Control of Dynamic*

Systems, Second Edition, Addison-Wesley, 1990.

[2] Van Loan, C.F., "Computing Integrals Involving the Matrix Exponential,"

IEEE Trans. Automatic Control, AC-15, October 1970.

See Also kalman Design Kalman estimator

lqgreg Assemble LQG regulator

1qrd Discrete LQ-optimal gain for continuous plant

Purpose

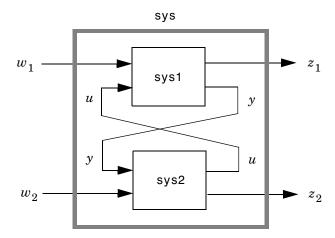
Redheffer star product (linear fractional transformation) of two LTI models

Syntax

Description

1ft forms the star product or linear fractional transformation (LFT) of two LTI models or LTI arrays. Such interconnections are widely used in robust control techniques.

sys = 1ft(sys1,sys2,nu,ny) forms the star product sys of the two LTI
models (or LTI arrays) sys1 and sys2. The star product amounts to the
following feedback connection for single LTI models (or for each model in an
LTI array).



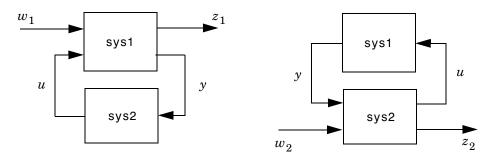
This feedback loop connects the first nu outputs of sys2 to the last nu inputs of sys1 (signals u), and the last ny outputs of sys1 to the first ny inputs of sys2 (signals y). The resulting system sys maps the input vector $[w_1; w_2]$ to the output vector $[z_1; z_2]$.

The abbreviated syntax

$$sys = lft(sys1, sys2)$$

produces:

- ullet The lower LFT of sys1 and sys2 if sys2 has fewer inputs and outputs than sys1. This amounts to deleting w_2 and z_2 in the above diagram.
- The upper LFT of sys1 and sys2 if sys1 has fewer inputs and outputs than sys2. This amounts to deleting w_1 and z_1 in the above diagram.



Lower LFT connection

Upper LFT connection

Algorithm

The closed-loop model is derived by elementary state-space manipulations.

Limitations

There should be no algebraic loop in the feedback connection.

See Also

connect Derive state-space model for block diagram interconnection

feedback Feedback connection

lqgreg

Purpose

Form LQG regulator given state-feedback gain and Kalman estimator

Syntax

```
rlqg = lqgreg(kest,k)
rlqg = lqgreg(kest,k,'current') % discrete-time only
rlqg = lqgreg(kest,k,controls)
```

Description

lqgreg forms the LQG regulator by connecting the Kalman estimator designed with kalman and the optimal state-feedback gain designed with lqr, dlqr, or lqry. The LQG regulator minimizes some quadratic cost function that trades off regulation performance and control effort. This regulator is dynamic and relies on noisy output measurements to generate the regulating commands.

In continuous time, the LQG regulator generates the commands

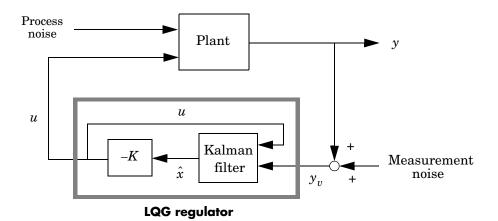
$$u = -K\hat{x}$$

where \hat{x} is the Kalman state estimate. The regulator state-space equations are

$$\hat{x} = \left[A - LC - (B - LD)K\right]\hat{x} + Ly_{v}$$

$$u = -K\hat{x}$$

where \boldsymbol{y}_v is the vector of plant output measurements (see kalman for background and notation). The diagram below shows this dynamic regulator in relation to the plant.



In discrete time, you can form the LQG regulator using either the prediction $\hat{x}[n|n-1]$ of x[n] based on measurements up to $y_v[n-1]$, or the current state estimate $\hat{x}[n|n]$ based on all available measurements including $y_v[n]$. While the regulator

$$u[n] = -K\hat{x}[n|n-1]$$

is always well-defined, the $\it current\ regulator$

$$u[n] = -K\hat{x}[n|n]$$

is causal only when I-KMD is invertible (see kalman for the notation). In addition, practical implementations of the current regulator should allow for the processing time required to compute u[n] once the measurements $y_v[n]$ become available (this amounts to a time delay in the feedback loop).

Usage

rlqg = lqgreg(kest,k) returns the LQG regulator rlqg (a state-space model) given the Kalman estimator kest and the state-feedback gain matrix k. The same function handles both continuous- and discrete-time cases. Use consistent tools to design kest and k:

- Continuous regulator for continuous plant: use lqr or lqry and kalman.
- Discrete regulator for discrete plant: use dlqr or lqry and kalman.

• Discrete regulator for continuous plant: use lqrd and kalmd.

In discrete time, lqgreg produces the regulator

$$u[n] = -K\hat{x}[n|n-1]$$

by default (see "Description"). To form the "current" LQG regulator instead, use

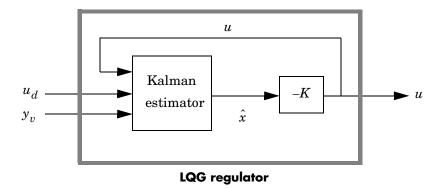
$$u[n] = -K\hat{x}[n|n]$$

the syntax

This syntax is meaningful only for discrete-time problems.

rlqg = lqgreg(kest,k,controls) handles estimators that have access to additional known plant inputs u_d . The index vector controls then specifies which estimator inputs are the controls u, and the resulting LQG regulator rlqg has u_d and y_v as inputs (see figure below).

Note Always use *positive* feedback to connect the LQG regulator to the plant.



Example

See the example LQG Regulation.

See Also	kalman	Kalman estimator design
----------	--------	-------------------------

kalmd Discrete Kalman estimator for continuous plant

lqr, dlqr State-feedback LQ regulator

1qrd Discrete LQ regulator for continuous plant

lqry LQ regulator with output weighting

reg Form regulator given state-feedback and estimator

gains

Purpose

Design linear-quadratic (LQ) state-feedback regulator for state-space systems

Syntax

Description

[K,S,e] = lqr(SYS,Q,R,N) calculates the optimal gain matrix K such that:

For a continuouse time system, the state-feedback law u = -Kx minimizes the quadratic cost function

$$J(u) = \int_0^\infty (x^T Q x + u^T R u + 2x^T N u) dt$$

subject to the system dynamics $\dot{x} = Ax + Bu$.

In addition to the state-feedback gain K, 1qr returns the solution S of the associated Riccati equation

$$A^{T}S + SA - (SB + N)R^{-1}(B^{T}S + N^{T}) + Q = 0$$

and the closed-loop eigenvalues e = eig(A-B*K). Note that K is derived from S by

$$K = R^{-1}(B^T S + N^T)$$

For a discrete-time state-space model, u[n]=-Kx[n] minimizes

$$J = \sum \{x'Qx + u'Ru + 2x'Nu\}$$

subject to x[n+1]=Ax[n]+Bu[n].

[K,S,e] = LQR(A,B,Q,R,N) is an equivalent syntax for continuous-time models with dynamics dx/dt=Ax+Bu.

In all cases, the default value N=0 is assumed when N is omitted.

Limitations

The problem data must satisfy:

- The pair (A, B) is stabilizable.
- R > 0 and $Q NR^{-1}N^T \ge 0$.

• $(Q - NR^{-1}N^T, A - BR^{-1}N^T)$ has no unobservable mode on the imaginary axis.

See Also

care Solve continuous Riccati equations

dlqr State-feedback LQ regulator for discrete plant

lqgreg Form LQG regulator

1qrd Discrete LQ regulator for continuous plant

1qry State-feedback LQ regulator with output weighting

lgrd

Purpose

Design discrete LQ regulator for continuous plant

Syntax

$$[Kd,S,e] = 1qrd(A,B,Q,R,Ts)$$

 $[Kd,S,e] = 1qrd(A,B,Q,R,N,Ts)$

Description

lqrd designs a discrete full-state-feedback regulator that has response characteristics similar to a continuous state-feedback regulator designed using lqr. This command is useful to design a gain matrix for digital implementation after a satisfactory continuous state-feedback gain has been designed.

[Kd,S,e] = lqrd(A,B,Q,R,Ts) calculates the discrete state-feedback law
$$u[n] = -K_d x[n]$$

that minimizes a discrete cost function equivalent to the continuous cost function

$$J = \int_0^\infty (x^T Q x + u^T R u) dt$$

The matrices A and B specify the continuous plant dynamics

$$\dot{x} = Ax + Bu$$

and Ts specifies the sample time of the discrete regulator. Also returned are the solution S of the discrete Riccati equation for the discretized problem and the discrete closed-loop eigenvalues e = eig(Ad-Bd*Kd).

[Kd,S,e] = 1qrd(A,B,Q,R,N,Ts) solves the more general problem with a cross-coupling term in the cost function.

$$J = \int_0^\infty (x^T Q x + u^T R u + 2x^T N u) dt$$

Algorithm

The equivalent discrete gain matrix Kd is determined by discretizing the continuous plant and weighting matrices using the sample time Ts and the zero-order hold approximation.

With the notation

$$\Phi(\tau) = e^{A\tau} \; , \qquad \qquad A_d = \Phi(T_s) \label{eq:delta_d$$

$$\Gamma(\tau) = \int_0^\tau e^{A\eta} B d\eta \ , \qquad B_d = \Gamma(T_s)$$

the discretized plant has equations

$$x[n+1] = A_d x[n] + B_d u[n]$$

and the weighting matrices for the equivalent discrete cost function are

$$\begin{bmatrix} Q_d & N_d \\ N_d^T & R_d \end{bmatrix} = \int_0^{T_s} \begin{bmatrix} \Phi^T(\tau) & 0 \\ \Gamma^T(\tau) & I \end{bmatrix} \begin{bmatrix} Q & N \\ N^T & R \end{bmatrix} \begin{bmatrix} \Phi(\tau) & \Gamma(\tau) \\ 0 & I \end{bmatrix} d\tau$$

The integrals are computed using matrix exponential formulas due to Van Loan (see [2]). The plant is discretized using c2d and the gain matrix is computed from the discretized data using d1qr.

Limitations

The discretized problem data should meet the requirements for dlgr.

References

- [1] Franklin, G.F., J.D. Powell, and M.L. Workman, *Digital Control of Dynamic Systems*, Second Edition, Addison-Wesley, 1980, pp. 439–440
- [2] Van Loan, C.F., "Computing Integrals Involving the Matrix Exponential," *IEEE Trans. Automatic Control*, AC-15, October 1970.

See Also

c2d	Discretization of LTI model
dlqr	State-feedback LQ regulator for discrete plant
kalmd	Discrete Kalman estimator for continuous plant
lqr	State-feedback LQ regulator for continuous plant $$

lqry

Purpose

Linear-quadratic (LQ) state-feedback regulator with output weighting

Syntax

Description

Given the plant

$$\dot{x} = Ax + Bu$$
$$y = Cx + Du$$

or its discrete-time counterpart, lqry designs a state-feedback control

$$u = -Kx$$

that minimizes the quadratic cost function with output weighting

$$J(u) = \int_0^\infty (y^T Q y + u^T R u + 2y^T N u) dt$$

(or its discrete-time counterpart). The function lqry is equivalent to lqr or dlqr with weighting matrices:

$$\begin{bmatrix} \overline{Q} & \overline{N} \\ \overline{N}^T & \overline{R} \end{bmatrix} = \begin{bmatrix} C^T & 0 \\ D^T & I \end{bmatrix} \begin{bmatrix} Q & N \\ N^T & R \end{bmatrix} \begin{bmatrix} C & D \\ 0 & I \end{bmatrix}$$

[K,S,e] = 1qry(sys,Q,R,N) returns the optimal gain matrix K, the Riccati solution S, and the closed-loop eigenvalues e = eig(A-B*K). The state-space model sys specifies the continuous- or discrete-time plant data (A,B,C,D). The default value N=0 is assumed when N is omitted.

Example

See LQG Design for the x-Axis for an example.

Limitations

The data $A, B, \overline{Q}, \overline{R}, \overline{N}$ must satisfy the requirements for lqr or dlqr.

See Also

 1qr
 State-feedback LQ regulator for continuous plant

 d1qr
 State-feedback LQ regulator for discrete plant

 $\begin{array}{ll} \hbox{\tt Kalman} & \hbox{\tt Kalman estimator design} \\ \hbox{\tt lqgreg} & \hbox{\tt Form LQG regulator} \end{array}$

Purpose

Simulate LTI model response to arbitrary inputs

Syntax

```
lsim(sys,u,t)
lsim(sys,u,t,x0)
lsim(sys,u,t,x0,'zoh')
lsim(sys,u,t,x0,'foh')

lsim(sys1,sys2,...,sysN,u,t)
lsim(sys1,sys2,...,sysN,u,t,x0)
lsim(sys1,'PlotStyle1',...,sysN,'PlotStyleN',u,t)
[y,t,x] = lsim(sys,u,t,x0)

lsim(sys)
```

Description

1sim simulates the (time) response of continuous or discrete linear systems to arbitrary inputs. When invoked without left-hand arguments, 1sim plots the response on the screen.

lsim(sys,u,t) produces a plot of the time response of the LTI model sys to the input time history t,u. The vector t specifies the time samples for the simulation and consists of regularly spaced time samples.

```
t = 0:dt:Tfinal
```

The matrix u must have as many rows as time samples (length(t)) and as many columns as system inputs. Each row u(i,:) specifies the input value(s) at the time sample t(i).

The LTI model sys can be continuous or discrete, SISO or MIMO. In discrete time, u must be sampled at the same rate as the system (t is then redundant and can be omitted or set to the empty matrix). In continuous time, the time sampling dt=t(2)-t(1) is used to discretize the continuous model. If dt is too large (undersampling), 1sim issues a warning suggesting that you use a more appropriate sample time, but will use the specified sample time. See Algorithm on page 155 for a discussion of sample times.

lsim(sys,u,t,x0) further specifies an initial condition x0 for the system states. This syntax applies only to state-space models.

lsim(sys,u,t,x0,'zoh') or lsim(sys,u,t,x0,'foh') explicitly specifies how the input values should be interpolated between samples (zero-order hold or linear interpolation). By default, lsim selects the interpolation method automatically based on the smoothness of the signal U.

Finally,

simulates the responses of several LTI models to the same input history t,u and plots these responses on a single figure. As with bode or plot, you can specify a particular color, linestyle, and/or marker for each system, for example,

The multisystem behavior is similar to that of bode or step.

When invoked with left-hand arguments,

```
[y,t] = lsim(sys,u,t)
[y,t,x] = lsim(sys,u,t) % for state-space models only
[y,t,x] = lsim(sys,u,t,x0) % with initial state
```

return the output response y, the time vector t used for simulation, and the state trajectories x (for state-space models only). No plot is drawn on the screen. The matrix y has as many rows as time samples (length(t)) and as many columns as system outputs. The same holds for x with "outputs" replaced by states. Note that the output t may differ from the specified time vector when the input data is undersampled (see Algorithm on page 155).

lsim(sys), with no additional arguments, opens the **Linear Simulation Tool**, which affords greater flexibility in specifying input signals and initial conditions. See "Specifying Input Signals and Input Conditions" in *Getting Started with the Control System Toolbox* for more information.

Example

Simulate and plot the response of the system

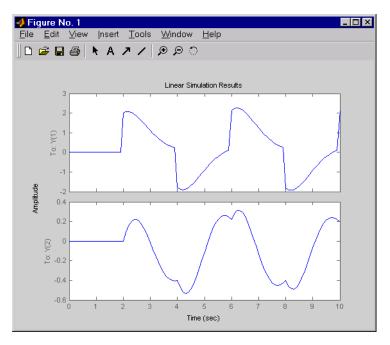
$$H(s) = \begin{bmatrix} \frac{2s^2 + 5s + 1}{s^2 + 2s + 3} \\ \frac{s - 1}{s^2 + s + 5} \end{bmatrix}$$

to a square wave with period of four seconds. First generate the square wave with gensig. Sample every 0.1 second during 10 seconds:

Then simulate with 1sim.

$$H = [tf([2 5 1],[1 2 3]) ; tf([1 -1],[1 1 5])]$$

 $lsim(H,u,t)$



Algorithm

Discrete-time systems are simulated with ltitr (state space) or filter (transfer function and zero-pole-gain).

Continuous-time systems are discretized with c2d using either the 'zoh' or 'foh' method ('foh' is used for smooth input signals and 'zoh' for discontinuous signals such as pulses or square waves). The sampling period is set to the spacing dt between the user-supplied time samples t.

The choice of sampling period can drastically affect simulation results. To illustrate why, consider the second-order model

$$H(s) = \frac{\omega^2}{s^2 + 2s + \omega^2}$$
, $\omega = 62.83$

To simulate its response to a square wave with period 1 second, you can proceed as follows:

```
w2 = 62.83^2

h = tf(w2,[1 \ 2 \ w2])

t = 0:0.1:5; % vector of time samples

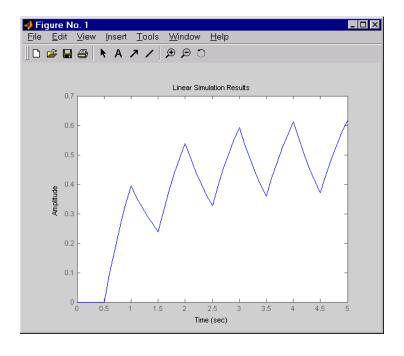
u = (rem(t,1)>=0.5); % square wave values

lsim(h,u,t)
```

1sim evaluates the specified sample time, gives this warning

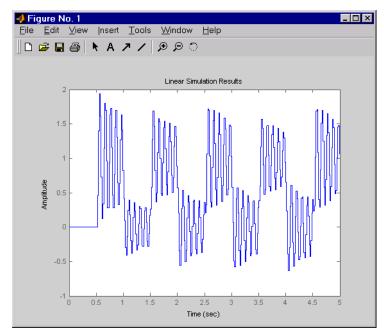
Warning: Input signal is undersampled. Sample every $0.016\ \text{sec}$ or faster.

and produces this plot.



To improve on this response, discretize H(s) using the recommended sampling period:

```
dt=0.016;
ts=0:dt:5;
us = (rem(ts,1)>=0.5)
hd = c2d(h,dt)
lsim(hd,us,ts)
```



This response exhibits strong oscillatory behavior hidden from the undersampled version.

See Also

gensig	Generate test input signals for 1sim
impulse	Impulse response
initial	Free response to initial condition
ltiview	LTI system viewer
step	Step response

Isimplot

Purpose

Simulate LTI model response to arbitrary inputs and return the plot handle

Syntax

```
h = lsimplot(sys)
h = lsimplot(sys1,sys2,...)
h = lsimplot(sys,u,t)
h = lsimplot(sys,u,t,x0)
h = lsimplot(sys1,sys2,...,u,t,x0)
h = lsimplot(AX,...)
h = lsimplot(..., plotoptions)
h = lsimplot(sys,u,t,x0,'zoh')
h = lsimplot(sys,u,t,x0,'foh')
```

Description

h = lsimplot(sys) opens the Linear Simulation Tool for the LTI model sys (created with tf, zpk, or ss), which enables interactive specification of driving input(s), the time vector, and initial state. It also returns the plot handle h. You can use this handle to customize the plot with the getoptions and setoptions commands. Type

```
help timeoptions
```

for a list of available plot options.

lsimplot(sys1, sys2,...) opens the Linear Simulation Tool for multiple LTI models sys1,sys2,.... Driving inputs are common to all specified systems but initial conditions can be specified separately for each.

lsimplot(sys,u,t) plots the time response of the LTI model sys to the input signal described by u and t. The time vector t consists of regularly spaced time samples. For MIMO systems, u is a matrix with as many columns as inputs and whose ith row specifies the input value at time t(i). For SISO systems u can be specified either as a row or column vector. For example,

```
t = 0:0.01:5;
u = sin(t);
lsimplot(sys,u,t)
```

simulates the response of a single-input model sys to the input u(t)=sin(t) during 5 seconds.

For discrete-time models, u should be sampled at the same rate as sys (t is then redundant and can be omitted or set to the empty matrix).

For continuous-time models, choose the sampling period t(2) - t(1) small enough to accurately describe the input u. 1sim issues a warning when u is undersampled, and hidden oscillations can occur.

lsimplot(sys,u,t,x0) specifies the initial state vector x0 at time t(1) (for state-space models only). x0 is set to zero when omitted.

lsimplot(sys1, sys2,...,u,t,x0) simulates the responses of multiple LTI models sys1,sys2,... on a single plot. The initial condition x0 is optional. You can also specify a color, line style, and marker for each system, as in

```
lsimplot(sys1,'r',sys2,'y--',sys3,'gx',u,t)
```

lsimplot(AX,...) plots into the axes with handle AX.

lsimplot(..., plotoptions) plots the initial condition response with the options specified in plotoptions. Type

help timeoptions

for more detail.

For continuous-time models, lsimplot(sys,u,t,x0,'zoh') or lsimplot(sys,u,t,x0,'foh') explicitly specifies how the input values should be interpolated between samples (zero-order hold or linear interpolation). By default, lsimplot selects the interpolation method automatically based on the smoothness of the signal u.

See Also

getoptions Get plot options

1sim Simulate LTI model response to arbitrary inputs

setoptions Set plot options

ltimodels

Purpose Help on LTI models

Syntax ltimodels

ltimodels(modeltype)

Description

ltimodels displays general information on the various types of LTI models supported in the Control System Toolbox.

ltimodels(modeltype) gives additional details and examples for each type of LTI model. The string modeltype selects the model type among the following:

- tf Transfer functions (TF objects)
- zpk Zero-pole-gain models (ZPK objects)
- ss State-space models (SS objects)
- frd Frequency response data models (FRD objects).

Note that you can type

ltimodels zpk

as a shorthand for

ltimodels('zpk')

See Also

frd Create or convert to FRD models
1tiprops Help on LTI model properties

ss Create or convert to a state-space model

tf Create or convert to a transfer function model zpk Create or convert to a zero/pole/gain model

Purpose Help on LTI model properties

Syntax ltimodels

ltimodels(modeltype)

Description ltiprops displays details on the generic properties of LTI models.

ltiprops(modeltype) gives details on the properties specific to the various
types of LTI models. The string modeltype selects the model type among the
following:

• tf — transfer functions (TF objects)

• zpk — zero-pole-gain models (ZPK objects)

• ss — state-space models (SS objects)

• frd — frequency response data (FRD objects).

Note that you can type

ltiprops tf

as a shorthand for

ltiprops('tf')

See also get Get the properties for an LTI model

ltimodels Help on LTI models

set Set or modify LTI model properties

Purpose

Initialize an LTI Viewer for LTI system response analysis

Syntax

```
ltiview
ltiview(sys1,sys2,...,sysn)
ltiview('plottype',sys1,sys2,...,sysn)
ltiview('plottype',sys,extras)
ltiview('clear',viewers)
ltiview('current',sys1,sys2,...,sysn,viewers)
```

Description

ltiview when invoked without input arguments, initializes a new LTI Viewer for LTI system response analysis.

ltiview(sys1,sys2,...,sysn) opens an LTI Viewer containing the step response of the LTI models sys1,sys2,...,sysn. You can specify a distinctive color, line style, and marker for each system, as in

```
sys1 = rss(3,2,2);
sys2 = rss(4,2,2);
ltiview(sys1,'r-*',sys2,'m--');
```

ltiview('plottype', sys) initializes an LTI Viewer containing the LTI response type indicated by plottype for the LTI model sys. The string plottype can be any one of the following:

```
'step'
'impulse'
'initial'
'lsim'
'pzmap'
'bode'
'nyquist'
'nichols'
```

or,

plottype can be a cell vector containing up to six of these plot types. For example,

```
ltiview({'step';'nyquist'},sys)
```

displays the plots of both of these response types for a given system sys.

ltiview(plottype, sys, extras) allows the additional input arguments supported by the various LTI model response functions to be passed to the ltiview command.

extras is one or more input arguments as specified by the function named in plottype. These arguments may be required or optional, depending on the type of LTI response. For example, if plottype is 'step' then extras may be the desired final time, Tfinal, as shown below.

```
ltiview('step',sys,Tfinal)
```

However, if *plottype* is 'initial', the extras arguments must contain the initial conditions x0 and may contain other arguments, such as Tfinal.

```
ltiview('initial',sys,x0,Tfinal)
```

See the individual references pages of each possible *plottype* commands for a list of appropriate arguments for extras.

ltiview('clear', viewers) clears the plots and data from the LTI Viewers with handles viewers.

ltiview('current',sys1,sys2,...,sysn,viewers) adds the responses of the systems sys1,sys2,...,sysn to the LTI Viewers with handles viewers. If these new systems do not have the same I/O dimensions as those currently in the LTI Viewer, the LTI Viewer is first cleared and only the new responses are shown.

Finally,

```
ltiview(plottype,sys1,sys2,...sysN)
ltiview(plottype,sys1,PlotStyle1,sys2,PlotStyle2,...)
ltiview(plottype,sys1,sys2,...sysN,extras)
```

initializes an LTI Viewer containing the responses of multiple LTI models, using the plot styles in PlotStyle, when applicable. See the individual reference pages of the LTI response functions for more information on specifying plot styles.

See Also

bode	Bode response
impulse	Impulse response
initial	Response to initial condition
lsim	Simulate LTI model response to arbitrary inputs

ltiview

nichols Nichols response nyquist Nyquist response pzmap Pole/zero map

sigma Singular value response

step Step response

Purpose

Solve continuous-time Lyapunov equations

Syntax

Description

lyap solves the special and general forms of the Lyapunov matrix equation. Lyapunov equations arise in several areas of control, including stability theory and the study of the RMS behavior of systems.

X = lyap(A,Q) solves the Lyapunov equation

$$AX + XA^T + Q = 0$$

where A and Q are square matrices of identical sizes. The solution X is a symmetric matrix if Q is.

X = lyap(A,B,C) solves the Sylvester equation

$$AX + XB + C = 0$$

The matrices A, B, and C must have compatible dimensions but need not be square.

X = lyap(A,Q,[],E) solves the generalized Lyapunov equation

$$AXE^T + EXA^T + Q = 0$$

where Q is a symmetric matrix. The empty square brackets, [], are mandatory. If you place any values inside them, the function will error out.

Algorithm

lyap transforms the A and B matrices to complex Schur form, computes the solution of the resulting triangular system, and transforms this solution back [1].

lyap uses SLICOT routines SB03MD and SG03AD for Lyapunov equations and SB04MD (SLICOT) and ZTRSYL (LAPACK) for Sylvester equations.

lyap

Limitations

The continuous Lyapunov equation has a (unique) solution if the eigenvalues $\alpha_1, \alpha_2, ..., \alpha_n$ of A and $\beta_1, \beta_2, ..., \beta_n$ of B satisfy

$$\alpha_i + \beta_i \neq 0$$
 for all pairs (i, j)

If this condition is violated, 1yap produces the error message

Solution does not exist or is not unique.

References

[1] Bartels, R.H. and G.W. Stewart, "Solution of the Matrix Equation AX + XB = C," *Comm. of the ACM*, Vol. 15, No. 9, 1972.

[2] Bryson, A.E. and Y.C. Ho, *Applied Optimal Control*, Hemisphere Publishing, 1975. pp. 328–338.

See Also

covar Covariance of system response to white noise

dlyap Solve discrete Lyapunov equations

Purpose Square-root solver for continuous-time Lyapunov equations

Syntax R = lypachol(A,B)

R = lyapchol(A,B,E)

DescriptionR = lyapchol(A,B) computes a Cholesky factorization X = R'*R of the solution X to the Lyapunov matrix equation:

A*X + X*A' + B*B' = 0

All eigenvalues of matrix A must lie in the open left half-plane for R to exist.

X = lyapchol(A,B,E) computes a Cholesky factorization X = R'*R of X solving the generalized Lyapunov equation:

A*X*E' + E*X*A' + B*B' = 0

All generalized eigenvalues of (A,E) must lie in the open left half-plane for R to exist.

Algorithm 1yapcho1 uses SLICOT routines SB03OD and SG03BD.

See Also lyap Solver for continuous-time Lyapunov equations

dlyapchol Square-root solver for discrete-time Lyapunov

equations

margin

Purpose

Compute gain and phase margins and associated crossover frequencies

Syntax

```
[Gm,Pm,Wcg,Wcp] = margin(sys)
[Gm,Pm,Wcg,Wcp] = margin(mag,phase,w)
margin(sys)
```

Description

margin calculates the minimum gain margin, phase margin, and associated crossover frequencies of SISO open-loop models. The gain and phase margins indicate the relative stability of the control system when the loop is closed. When invoked without left-hand arguments, margin produces a Bode plot and displays the margins on this plot.

The gain margin is the amount of gain increase required to make the loop gain unity at the frequency where the phase angle is -180° . In other words, the gain margin is 1/g if g is the gain at the -180° phase frequency. Similarly, the phase margin is the difference between the phase of the response and -180° when the loop gain is 1.0. The frequency at which the magnitude is 1.0 is called the unity-gain frequency or crossover frequency. It is generally found that gain margins of three or more combined with phase margins between 30 and 60 degrees result in reasonable trade-offs between bandwidth and stability.

[Gm,Pm,Wcg,Wcp] = margin(sys) computes the gain margin Gm, the phase margin Pm, and the corresponding crossover frequencies Wcg and Wcp, given the SISO open-loop model sys. This function handles both continuous- and discrete-time cases. When faced with several crossover frequencies, margin returns the smallest gain and phase margins.

[Gm,Pm,Wcg,Wcp] = margin(mag,phase,w) derives the gain and phase margins from the Bode frequency response data (magnitude, phase, and frequency vector). Interpolation is performed between the frequency points to estimate the margin values. This approach is generally less accurate.

When invoked without left-hand argument,

```
margin(sys)
```

plots the open-loop Bode response with the gain and phase margins marked by vertical lines.

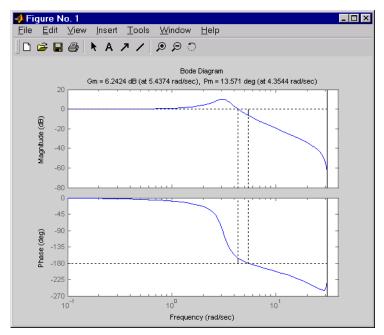
Example

You can compute the gain and phase margins of the open-loop discrete-time transfer function. Type $\,$

You can also display these margins graphically.

margin





Algorithm

The phase margin is computed using H_{∞} theory, and the gain margin by solving $H(j\omega)=\overline{H(j\omega)}$ for the frequency ω .

See Also

bode Bode frequency response ltiview LTI system viewer

Purpose

Minimal realization or pole-zero cancellation

Syntax

```
sysr = minreal(sys)
sysr = minreal(sys,tol)
[sysr,u] = minreal(sys,tol)
```

Description

sysr = minreal(sys) eliminates uncontrollable or unobservable state in state-space models, or cancels pole-zero pairs in transfer functions or zero-pole-gain models. The output sysr has minimal order and the same response characteristics as the original model sys.

sysr = minreal(sys,tol) specifies the tolerance used for state elimination or
pole-zero cancellation. The default value is tol = sqrt(eps) and increasing
this tolerance forces additional cancellations.

[sysr,u] = minreal(sys,to1) returns, for state-space model sys, an orthogonal matrix U such that (U*A*U',U*B,C*U') is a Kalman decomposition of (A,B,C)

Example

The commands

```
g = zpk([],1,1)
h = tf([2 1],[1 0])
cloop = inv(1+g*h) * g
```

produce the nonminimal zero-pole-gain model by typing cloop.

```
Zero/pole/gain:
s (s-1)
```

(s-1) $(s^2 + s + 1)$

To cancel the pole-zero pair at s = 1, type

```
cloop = minreal(cloop)
```

and MATLAB returns

Zero/pole/gain:

```
(s^2 + s + 1)
```

minreal

Algorithm Pole-zero cancellation is a straightforward search through the poles and zeros

looking for matches that are within tolerance. Transfer functions are first

converted to zero-pole-gain form.

See Also balreal Grammian-based input/output balancing

modred Model order reduction

sminreal Structured model reduction

Purpose

Model order reduction

Syntax

```
rsys = modred(sys,elim)
rsys = modred(sys,elim,'method')
```

Description

modred reduces the order of a continuous or discrete state-space model sys by eliminating the states found in the vector elim. The full state vector X is partitioned as X = [X1;X2] where X2 is to be discarded, and the reduced state is set to Xr = X1+T*X2 where T is chosen to enforce matching DC gains (steady-state response) between sys and rsys.

elim can be a vector of indices or a logical vector commensurate with X where true values mark states to be discarded. This function is usually used in conjunction with balreal. Use balreal to first isolate states with negligible contribution to the I/O response. If sys has been balanced with balreal and the vector g of Hankel singular values has M small entries, you can use modred to eliminate the corresponding M states. For example:

```
[sys,g] = balreal(sys) % Compute balanced realization
elim = (g<1e-8) % Small entries of g are negligible states
rsys = modred(sys,elim) % Remove negligible states</pre>
```

rsys = modred(sys,elim,'method') also specifies the state elimination
method. Choices for 'method' include

- 'MatchDC': Enforce matching DC gains (default)
- 'Truncate': Simply delete X2 and sets Xr = X1.

The 'Truncate' option tends to produces a better approximation in the frequency domain, but the DC gains are not guaranteed to match.

If the state-space model sys has been balanced with balreal and the grammians have m small diagonal entries, you can reduce the model order by eliminating the last m states with modred.

Example 1

Consider the continuous fourth-order model

$$h(s) = \frac{s^3 + 11s^2 + 36s + 26}{s^4 + 14.6s^3 + 74.96s^2 + 153.7s + 99.65}$$

To reduce its order, first compute a balanced state-space realization with balreal by typing

```
h = tf([1 11 36 26],[1 14.6 74.96 153.7 99.65])
[hb,g] = balreal(h)
g'
```

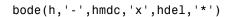
MATLAB returns

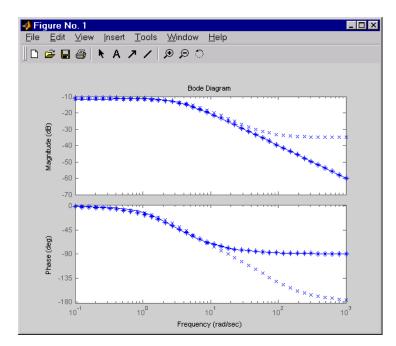
```
ans = 1.3938e-01 9.5482e-03 6.2712e-04 7.3245e-06
```

The last three diagonal entries of the balanced grammians are small, so eliminate the last three states with modred using both matched DC gain and direct deletion methods.

```
hmdc = modred(hb,2:4,'MatchDC')
hdel = modred(hb,2:4,'Truncate')
```

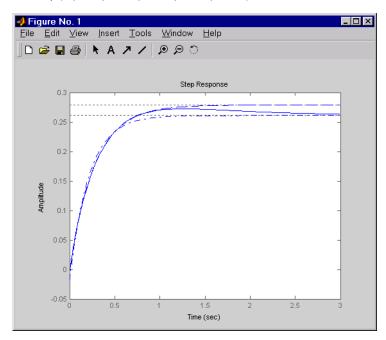
Both hmdc and hdel are first-order models. Compare their Bode responses against that of the original model h(s).





The reduced-order model hdel is clearly a better frequency-domain approximation of h(s) . Now compare the step responses.

step(h,'-',hmdc,'-.',hdel,'--')



While hdel accurately reflects the transient behavior, only hmdc gives the true steady-state response.

Algorithm

The algorithm for the matched DC gain method is as follows. For continuous-time models

$$\dot{x} = Ax + Bu$$

$$y = Cx + Du$$

the state vector is partitioned into \boldsymbol{x}_1 , to be kept, and \boldsymbol{x}_2 , to be eliminated.

$$\begin{bmatrix} \dot{x}_1 \\ \dot{x}_2 \end{bmatrix} = \begin{bmatrix} A_{11} & A_{12} \\ A_{21} & A_{22} \end{bmatrix} \begin{bmatrix} x_1 \\ x_2 \end{bmatrix} + \begin{bmatrix} B_1 \\ B_2 \end{bmatrix} u$$
$$y = \begin{bmatrix} C_1 & C_2 \end{bmatrix} x + Du$$

Next, the derivative of x_2 is set to zero and the resulting equation is solved for x_1 . The reduced-order model is given by

$$\begin{aligned} x_1 &= [A_{11} - A_{12} A_{22}^{-1} A_{21}] x_1 + [B_1 - A_{12} A_{22}^{-1} B_2] u \\ y &= [C_1 - C_2 A_{22}^{-1} A_{21}] x + [D - C_2 A_{22}^{-1} B_2] u \end{aligned}$$

The discrete-time case is treated similarly by setting

$$x_2[n+1] = x_2[n]$$

Limitations

With the matched DC gain method, A_{22} must be invertible in continuous time, and $I-A_{22}$ must be invertible in discrete time.

See Also

balreal Input/output balancing of state-space models
minreal Minimal state-space realizations

Region-based modal decomposition

Syntax

```
[H,H0] = modsep(G,N,REGIONFCN)
[H,H0] = modsep(G,N,REGIONFCN,PARAM1,...)
```

Description

[H,H0] = modsep(G,N,REGIONFCN) decomposes the LTI model G into a sum of n simpler models Hj with their poles in disjoint regions Rj of the complex plane:

$$G(s) = H0 + \sum_{j=1}^{N} Hj(s)$$

G can be any LTI model created with ss, tf, or zpk, and N is the number of regions used in the decomposition. modsep packs the submodels Hj into an LTI array H and returns the static gain HO separately. Use H(:,:,j) to retrieve the submodel Hj(s).

To specify the regions of interest, use a function of the form

```
IR = REGIONFCN(p)
```

that assigns a region index IR between 1 and N to a given pole p. You can specify this function as a string or a function handle, and use the syntax MODSEP(G,N,REGIONFCN,PARAM1,...) to pass extra input arguments:

```
IR = REGIONFCN(p,PARAM1,...)
```

Example

To decompose G into G(z) = H0 + H1(z) + H2(z) where H1 and H2 have their poles inside and outside the unit disk respectively, use

```
[H,H0] = modsep(G,2,@udsep)
```

where the function udsep is defined by

```
function r = udsep(p)
if abs(p)<1, r = 1; % assign r=1 to poles inside unit disk
else r = 2; % assign r=2 to poles outside unit disk
end
```

To extract H1(z) and H2(z) from the LTI array H, use

$$H1 = H(:,:,1); H2 = H(:,:,2);$$

See Also

stabsep

Stable/unstable decomposition of LTI models

Provide the number of the dimensions of an LTI model or LTI array

Syntax

n = ndims(sys)

Description

n = ndims(sys) is the number of dimensions of an LTI model or an array of LTI models sys. A single LTI model has two dimensions (one for outputs, and one for inputs). An LTI array has 2+p dimensions, where $p \ge 2$ is the number of array dimensions. For example, a 2-by-3-by-4 array of models has 2+3=5 dimensions.

```
ndims(sys) = length(size(sys))
```

Example

ndims returns 4 for this 3-by-1 array of SISO models.

See Also

size

Returns a vector containing the lengths of the dimensions of an LTI array or model

Superimpose a Nichols chart on a Nichols plot

Syntax

ngrid

Description

ngrid superimposes Nichols chart grid lines over the Nichols frequency response of a SISO LTI system. The range of the Nichols grid lines is set to encompass the entire Nichols frequency response.

The chart relates the complex number H/(1+H) to H, where H is any complex number. For SISO systems, when H is a point on the open-loop frequency response, then

$$\frac{H}{1+H}$$

is the corresponding value of the closed-loop frequency response assuming unit negative feedback.

If the current axis is empty, ngrid generates a new Nichols chart grid in the region -40 dB to 40 dB in magnitude and -360 degrees to 0 degrees in phase. If the current axis does not contain a SISO Nichols frequency response, ngrid returns a warning.

Example

Plot the Nichols response with Nichols grid lines for the system.

$$H(s) = \frac{-4s^4 + 48s^3 - 18s^2 + 250s + 600}{s^4 + 30s^3 + 282s^2 + 525s + 60}$$

Type

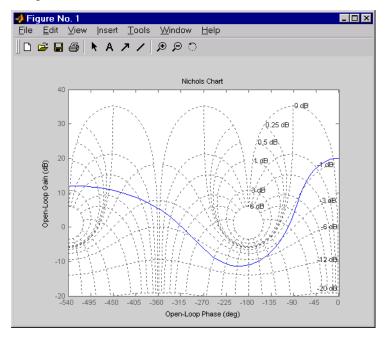
$$H = tf([-4 \ 48 \ -18 \ 250 \ 600],[1 \ 30 \ 282 \ 525 \ 60])$$

MATLAB returns

Type

nichols(H)

ngrid



See Also nichols Nichols plots

Compute Nichols frequency response of LTI models

Syntax

```
nichols(sys)
nichols(sys,w)

nichols(sys1,sys2,...,sysN)
nichols(sys1,sys2,...,sysN,w)
nichols(sys1,'PlotStyle1',...,sysN,'PlotStyleN')

[mag,phase,w] = nichols(sys)
[mag,phase] = nichols(sys,w)
```

Description

nichols computes the frequency response of an LTI model and plots it in the Nichols coordinates. Nichols plots are useful to analyze open- and closed-loop properties of SISO systems, but offer little insight into MIMO control loops. Use ngrid to superimpose a Nichols chart on an existing SISO Nichols plot.

nichols (sys) produces a Nichols plot of the LTI model sys. This model can be continuous or discrete, SISO or MIMO. In the MIMO case, nichols produces an array of Nichols plots, each plot showing the response of one particular I/O channel. The frequency range and gridding are determined automatically based on the system poles and zeros.

nichols(sys,w) explicitly specifies the frequency range or frequency points to be used for the plot. To focus on a particular frequency interval [wmin,wmax], set w = {wmin,wmax}. To use particular frequency points, set w to the vector of desired frequencies. Use logspace to generate logarithmically spaced frequency vectors. Frequencies should be specified in radians/sec.

nichols(sys1,sys2,...,sysN) or nichols(sys1,sys2,...,sysN,w) superimposes the Nichols plots of several LTI models on a single figure. All systems must have the same number of inputs and outputs, but may otherwise be a mix of continuous- and discrete-time systems. You can also specify a distinctive color, linestyle, and/or marker for each system plot with the syntax

```
nichols(sys1,'PlotStyle1',...,sysN,'PlotStyleN')
```

See bode for an example.

When invoked with left-hand arguments,

```
[mag,phase,w] = nichols(sys)
[mag,phase] = nichols(sys,w)
```

return the magnitude and phase (in degrees) of the frequency response at the frequencies w (in rad/sec). The outputs mag and phase are 3-D arrays similar to those produced by bode (see the bode reference page). They have dimensions

 $(number of outputs) \times (number of inputs) \times (length of w)$

Example

Plot the Nichols response of the system

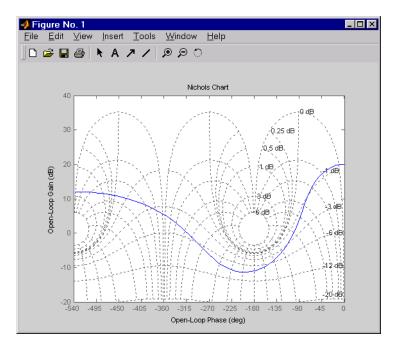
$$H(s) = \frac{-4s^4 + 48s^3 - 18s^2 + 250s + 600}{s^4 + 30s^3 + 282s^2 + 525s + 60}$$

$$num = [-4 \ 48 \ -18 \ 250 \ 600];$$

$$den = [1 \ 30 \ 282 \ 525 \ 60];$$

$$H = tf(num, den)$$

$$nichols(H); ngrid$$



The right-click menu for Nichols plots includes the **Tight** option under **Zoom**. You can use this to clip unbounded branches of the Nichols plot.

Algorithm

See bode.

See Also

bode	Bode plot	
evalfr	Response at single complex frequency	
freqresp	Frequency response computation	
ltiview	LTI system viewer	
ngrid	Grid on Nichols plot	
nyquist	Nyquist plot	
sigma	Singular value plot	

Plot Nichols frequency response and return the plot handle

Syntax

```
h = nicholsplot(sys)
```

```
h = nicholsplot(sys,{wmin,wmax})
```

h = nicholsplot(sys,w)

h = nicholsplot(sys1,sys2,...,w)

h = nicholsplot(AX,...)

h = nicholsplot(..., plotoptions)

Description

h = nicholsplot(sys) draws the nichols plot of the LTI model sys (created with tf, zpk, ss, or frd). It also returns the plot handle h. You can use this handle to customize the plot with the getoptions and setoptions commands. Type

help nicholsoptions

for a list of available plot options.

The frequency range and number of points are chosen automatically. See bode for details on the notion of frequency in discrete time.

nicholsplot(sys, {wmin, wmax}) draws the Nichols plot for frequencies between wmin and wmax (in rad/s).

nicholsplot(sys,w) uses the user-supplied vector w of frequencies, in radians/second, at which the Nichols response is to be evaluated. See logspace to generate logarithmically spaced frequency vectors.

nicholsplot(sys1,sys2,...,w) draws the Nichols plots of multiple LTI models sys1,sys2,... on a single plot. The frequency vector w is optional. You can also specify a color, line style, and marker for each system, as in

```
nicholsplot(sys1,'r',sys2,'y--',sys3,'gx').
```

nicholsplot(AX,...) plots into the axes with handle AX.

nicholsplot(..., plotoptions) plots the Nichols plot with the options specified in plotoptions. Type

help nicholsoptions

for more details.

nicholsplot

Example

Generate a Nichols plot and use the plot handle to change the frequency units to Hz.

```
sys = rss(5);
h = nicholsplot(sys);
% Change units to Hz
setoptions(h,'FreqUnits','Hz');
```

See Also

getoptions Get plot options

nichols Compute Nichols frequency responses of LTI models

setoptions Set plot options

Compute LTI model norms

Syntax

```
norm(sys)
norm(sys,2)

norm(sys,inf)
norm(sys,inf,tol)
[ninf,fpeak] = norm(sys)
```

Description

norm computes the H_2 or L_∞ norm of a continuous- or discrete-time LTI model.

H₂ Norm

The H_2 norm of a stable continuous system with transfer function H(s), is the root-mean-square of its impulse response, or equivalently

$$\|H\|_{2} = \sqrt{\frac{1}{2\pi} \int_{-\infty}^{\infty} \operatorname{Trace}(H(j\omega)^{H} H(j\omega)) \ d\omega}$$

This norm measures the steady-state covariance (or power) of the output response y = Hw to unit white noise inputs w.

$$||H||_{2}^{2} = \lim_{t \to \infty} E\{y(t)^{T}y(t)\}, \qquad E(w(t)w(\tau)^{T}) = \delta(t-\tau)I$$

Infinity Norm

The infinity norm is the peak gain of the frequency response, that is,

$$\|H(s)\|_{\infty} = \max_{\omega} |H(j\omega)|$$
 (SISO case)

$$\|H(s)\|_{\infty} = \max_{\omega} \sigma_{\max}(H(j\omega))$$
 (MIMO case)

where $\sigma_{max}(.)\,$ denotes the largest singular value of a matrix.

The discrete-time counterpart is

$$\|H(z)\|_{\infty} = \max_{\theta \in [0, \pi]} \sigma_{\max}(H(e^{j\theta}))$$

Usage

norm(sys) or norm(sys,2) both return the H_2 norm of the TF, SS, or ZPK model sys. This norm is infinite in the following cases:

- sys is unstable.
- sys is continuous and has a nonzero feedthrough (that is, nonzero gain at the frequency $\omega = \infty$).

Note that norm(sys) produces the same result as

```
sqrt(trace(covar(sys,1)))
```

norm(sys,inf) computes the infinity norm of any type of LTI model sys. This norm is infinite if sys has poles on the imaginary axis in continuous time, or on the unit circle in discrete time.

norm(sys,inf,tol) sets the desired relative accuracy on the computed infinity norm (the default value is tol=1e-2).

[ninf,fpeak] = norm(sys,inf) also returns the frequency fpeak where the
gain achieves its peak value.

Example

Consider the discrete-time transfer function

$$H(z) = \frac{z^3 - 2.841z^2 + 2.875z - 1.004}{z^3 - 2.417z^2 + 2.003z - 0.5488}$$

with sample time 0.1 second. Compute its ${\cal H}_2$ norm by typing

Compute its infinity norm by typing

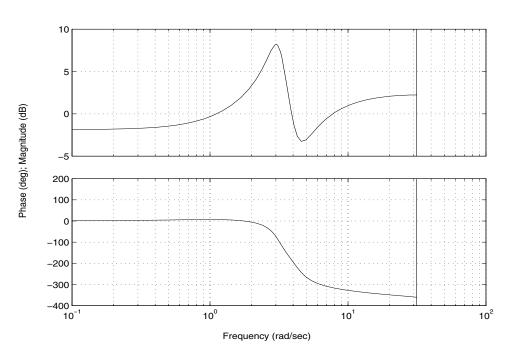
1.2438

$$[ninf,fpeak] = norm(H,inf)$$

These values are confirmed by the Bode plot of H(z).

bode(H)





The gain indeed peaks at approximately 3 rad/sec and its peak value in dB is found by typing

20*log10(ninf)
MATLAB returns
ans =

norm

8.1268

Algorithm norm uses the same algorithm as covar for the H_2 norm, and the algorithm of

[1] for the infinity norm. sys is first converted to state space.

References [1] Bruisma, N.A. and M. Steinbuch, "A Fast Algorithm to Compute the

 H_{∞} -Norm of a Transfer Function Matrix," System Control Letters, 14 (1990),

pp. 287-293.

See Also bode Bode plot

frequency response computation

sigma Singular value plot

Compute Nyquist frequency response of LTI models

Syntax

```
nyquist(sys)
nyquist(sys,w)

nyquist(sys1,sys2,...,sysN)
nyquist(sys1,sys2,...,sysN,w)
nyquist(sys1,'PlotStyle1',...,sysN,'PlotStyleN')

[re,im,w] = nyquist(sys)
[re,im] = nyquist(sys,w)
```

Description

nyquist calculates the Nyquist frequency response of LTI models. When invoked without left-hand arguments, nyquist produces a Nyquist plot on the screen. Nyquist plots are used to analyze system properties including gain margin, phase margin, and stability.

nyquist (sys) plots the Nyquist response of an arbitrary LTI model sys. This model can be continuous or discrete, and SISO or MIMO. In the MIMO case, nyquist produces an array of Nyquist plots, each plot showing the response of one particular I/O channel. The frequency points are chosen automatically based on the system poles and zeros.

nyquist(sys,w) explicitly specifies the frequency range or frequency points to be used for the plot. To focus on a particular frequency interval, set
 w = {wmin,wmax}. To use particular frequency points, set w to the vector of desired frequencies. Use logspace to generate logarithmically spaced frequency vectors. Frequencies should be specified in rad/sec.

nyquist(sys1,sys2,...,sysN) or nyquist(sys1,sys2,...,sysN,w) superimposes the Nyquist plots of several LTI models on a single figure. All systems must have the same number of inputs and outputs, but may otherwise be a mix of continuous- and discrete-time systems. You can also specify a distinctive color, linestyle, and/or marker for each system plot with the syntax

```
nyquist(sys1,'PlotStyle1',...,sysN,'PlotStyleN')
```

See bode for an example.

When invoked with left-hand arguments

```
[re,im,w] = nyquist(sys)
[re,im] = nyquist(sys,w)
```

return the real and imaginary parts of the frequency response at the frequencies w (in rad/sec). re and im are 3-D arrays (see "Arguments" below for details).

Arguments

The output arguments re and im are 3-D arrays with dimensions

$$(number of outputs) \times (number of inputs) \times (length of w)$$

For SISO systems, the scalars re(1,1,k) and im(1,1,k) are the real and imaginary parts of the response at the frequency $\omega_k = \mathrm{w}(k)$.

$$\begin{split} \operatorname{re}(1,&1,&\mathbf{k}) = \operatorname{Re}(h(j\omega_k)) \\ &\operatorname{im}(1,&1,&\mathbf{k}) = \operatorname{Im}(h(j\omega_k)) \end{split}$$

For MIMO systems with transfer function H(s), re(:,:,k) and im(:,:,k) give the real and imaginary parts of $H(j\omega_k)$ (both arrays with as many rows as outputs and as many columns as inputs). Thus,

$$\begin{aligned} & \operatorname{re}(\mathbf{i},\mathbf{j},\mathbf{k}) = \operatorname{Re}(h_{ij}(j\omega_k)) \\ & \operatorname{im}(\mathbf{i},\mathbf{j},\mathbf{k}) = \operatorname{Im}(h_{ij}(j\omega_k)) \end{aligned}$$

where \boldsymbol{h}_{ij} is the transfer function from input j to output i .

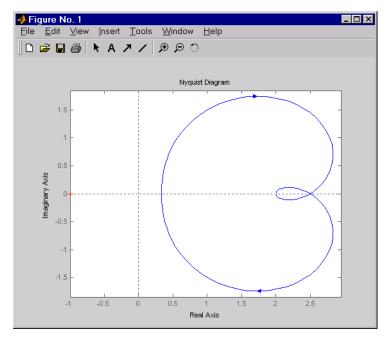
Example

Plot the Nyquist response of the system

$$H(s) = \frac{2s^2 + 5s + 1}{s^2 + 2s + 3}$$

$$H = tf([2 5 1],[1 2 3])$$





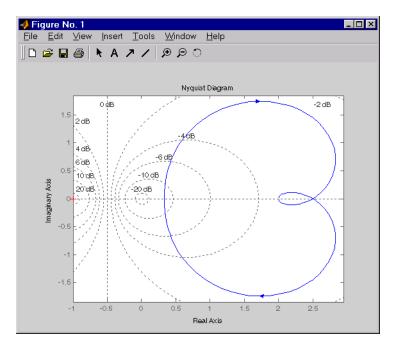
The nyquist function has support for M-circles, which are the contours of the constant closed-loop magnitude. M-circles are defined as the locus of complex numbers where

$$T(j\omega) = \left| \frac{G(j\omega)}{1 + G(j\omega)} \right|$$

is a constant value. In this equation, ω is the frequency in radians/second, and G is the collection of complex numbers that satisfy the constant magnitude requirement.

To activate the grid, select **Grid** from the right-click menu or type grid

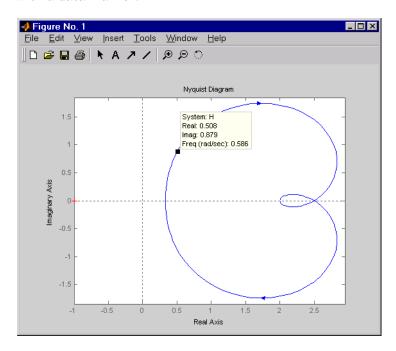
at the MATLAB prompt. This figure shows the M circles for transfer function ${\cal H}.$



You have two zoom options available from the right-click menu that apply specifically to Nyquist plots:

- \bullet Tight —Clips unbounded branches of the Nyquist plot, but still includes the critical point (-1, 0)
- On (-1,0) Zooms around the critical point (-1,0)

Also, click anywhere on the curve to activate data markers that display the real and imaginary values at a given frequency. This figure shows the nyquist plot with a data marker.



See Also

bode	Bode plot	
evalfr	Response at single complex frequency	
freqresp	Frequency response computation	
ltiview	LTI system viewer	
nichols	Nichols plot	
sigma	Singular value plot	

nyquistplot

Purpose

Plot Nyquist frequency response and return the plot handle

Syntax

```
h = nyquistplot(sys)
```

h = nyquistplot(sys,{wmin,wmax})

h = nyquistplot(sys,w)

h = nyquistplot(sys1,sys2,...,w)

h = nyquistplot(AX,...)

h = nyquistplot(..., plotoptions)

Description

h = nyquistplot(sys) draws the Nyquist plot of the LTI model sys (created with tf, zpk, ss, or frd). It also returns the plot handle h. You can use this handle to customize the plot with the getoptions and setoptions commands. Type

help nyquistoptions

for a list of available plot options.

The frequency range and number of points are chosen automatically. See bode for details on the notion of frequency in discrete time.

nyquistplot(sys, {wmin, wmax}) draws the Nyquist plot for frequencies between wmin and wmax (in rad/s).

nyquistplot(sys,w) uses the user-supplied vector w of frequencies (in rad/s) at which the Nyquist response is to be evaluated. See logspace to generate logarithmically spaced frequency vectors.

nyquistplot(sys1,sys2,...,w) draws the Nyquist plots of multiple LTI models sys1,sys2,... on a single plot. The frequency vector w is optional. You can also specify a color, line style, and marker for each system, as in

```
nyquistplot(sys1,'r',sys2,'y--',sys3,'gx').
```

nyquistplot(AX,...) plots into the axes with handle AX.

nyquistplot(..., plotoptions) plots the Nyquist response with the options specified in plotoptions. Type

help nyquistoptions

for more details.

Example Plot the Nyquist frequency response and change the units to Hz.

```
sys = rss(5);
h = nicholsplot(sys);
% Change units to Hz
setoptions(h,'FreqUnits','Hz');
```

See Also getoptions Get plot options

nyquist Nyquist frequency response of LTI models

setoptions Set plot options

Form the observability matrix

Syntax

Description

obsv computes the observability matrix for state-space systems. For an n-by-n matrix A and a p-by-n matrix C, obsv(A,C) returns the observability matrix

$$Ob = \begin{bmatrix} C \\ CA \\ CA^2 \\ \vdots \\ CA^{n-1} \end{bmatrix}$$

with n columns and np rows.

Ob = obsv(sys) calculates the observability matrix of the state-space model sys. This syntax is equivalent to executing

The model is observable if 0b has full rank n.

Example

Determine if the pair

is observable. Type

$$Ob = obsv(A,C);$$

% Number of unobservable states
unob = length(A)-rank(Ob)

MATLAB responds with

unob = 0

See Also

obsvf

Compute the observability staircase form

Compute the observability staircase form

Syntax

Description

If the observability matrix of (A,C) has rank $r \le n$, where n is the size of A, then there exists a similarity transformation such that

$$\overline{A} = TAT^T$$
, $\overline{B} = TB$, $\overline{C} = CT^T$

where *T* is unitary and the transformed system has a *staircase* form with the unobservable modes, if any, in the upper left corner.

$$\overline{A} = \begin{bmatrix} A_{no} & A_{12} \\ 0 & A_o \end{bmatrix}, \qquad \overline{B} = \begin{bmatrix} B_{no} \\ B_o \end{bmatrix}, \qquad \overline{C} = \begin{bmatrix} 0 & C_o \end{bmatrix}$$

where (C_o, A_o) is observable, and the eigenvalues of A_{no} are the unobservable modes.

[Abar,Bbar,Cbar,T,k] = obsvf(A,B,C) decomposes the state-space system with matrices A, B, and C into the observability staircase form Abar, Bbar, and Cbar, as described above. T is the similarity transformation matrix and k is a vector of length n, where n is the number of states in A. Each entry of k represents the number of observable states factored out during each step of the transformation matrix calculation [1]. The number of nonzero elements in k indicates how many iterations were necessary to calculate T, and sum(k) is the number of states in A_n , the observable portion of Abar.

obsvf(A,B,C,tol) uses the tolerance tol when calculating the observable/unobservable subspaces. When the tolerance is not specified, it defaults to 10*n*norm(a,1)*eps.

Example

Form the observability staircase form of

B =

by typing

[Abar,Bbar,Cbar,T,k] = obsvf(A,B,C)

Algorithm

obsvf is an M-file that implements the Staircase Algorithm of [1] by calling ctrbf and using duality.

References

[1] Rosenbrock, M.M., $State\mbox{-}Space$ and Multivariable Theory, John Wiley, 1970.

See Also

ctrbf Compute the controllability staircase form obsv Calculate the observability matrix

Generate continuous second-order systems

Syntax

$$[A,B,C,D] = ord2(wn,z)$$

 $[num,den] = ord2(wn,z)$

Description

[A,B,C,D] = ord2(wn,z) generates the state-space description (A,B,C,D) of the second-order system

$$h(s) = \frac{1}{s^2 + 2\zeta \omega_n s + \omega_n^2}$$

given the natural frequency wn (ω_n) and damping factor $z(\zeta)$. Use ss to turn this description into a state-space object.

[num,den] = ord2(wn,z) returns the numerator and denominator of the second-order transfer function. Use tf to form the corresponding transfer function object.

Example

To generate an LTI model of the second-order transfer function with damping factor $\zeta=0.4$ and natural frequency $\omega_n=2.4$ rad/sec., type

See Also

rss Generate random stable continuous models
ss Create a state-space LTI model
tf Create a transfer function LTI model

Purpose LTI model order

Description ns = order(sys) returns the model order ns. The order of an LTI model is the

number of poles (for transfer functions) or the number of states (for state-space

models).

For LTI arrays, ns is an array of the same size listing the orders of each model

in sys.

order(sys) is an overloaded method that accepts SS, TF, and ZPK models.

See Also pole Computes the poles of LTI models

 $\begin{array}{ll} \mbox{balred} & \mbox{Model order reduction} \\ \mbox{ltimodels} & \mbox{Help on LTI models} \end{array}$

Compute the Padé approximation of models with time delays

Syntax

```
[num,den] = pade(T,N)
pade(T,N)

sysx = pade(sys,N)
sysx = pade(sys,NI,NO,Nio)
```

Description

pade approximates time delays by rational LTI models. Such approximations are useful to model time delay effects such as transport and computation delays within the context of continuous-time systems. The Laplace transform of an time delay of T seconds is $\exp(-sT)$. This exponential transfer function is approximated by a rational transfer function using the Padé approximation formulas [1].

[num,den] = pade(T,N) returns the Nth-order (diagonal) Padé approximation of the continuous-time I/O delay $\exp(-sT)$ in transfer function form. The row vectors num and den contain the numerator and denominator coefficients in descending powers of s. Both are Nth-order polynomials.

When invoked without output arguments,

```
pade(T,N)
```

plots the step and phase responses of the Nth-order Padé approximation and compares them with the exact responses of the model with I/O delay T. Note that the Padé approximation has unit gain at all frequencies.

sysx = pade(sys,N) produces a delay-free approximation sysx of the continuous delay system sys. All delays are replaced by their Nth-order Padé approximation. See Time Delays for details on LTI models with delays.

sysx = pade(sys,NI,NO,Nio) specifies independent approximation orders for each input, output, and I/O delay. These approximation orders are given by the arrays of integers NI, NO, and Nio, such that:

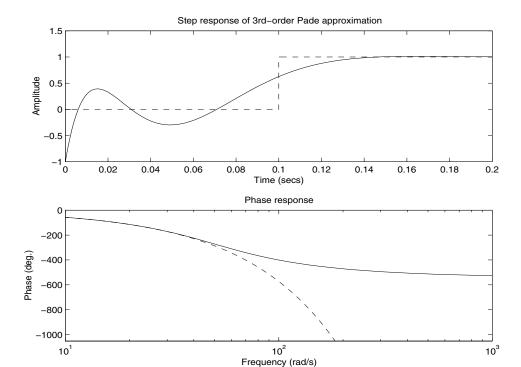
- NI(j) is the approximation order for the j-th input channel.
- NO(i) is the approximation order for the i-th output channel.
- Nio(i,j) is the approximation order for the I/O delay from input j to output i.

You can use scalar values to specify uniform approximation orders, and [] if there are no input, output, or I/O delays.

Example

Compute a third-order Padé approximation of a 0.1 second I/O delay and compare the time and frequency responses of the true delay and its approximation. To do this, type

pade(0.1,3)



Limitations

High-order Padé approximations produce transfer functions with clustered poles. Because such pole configurations tend to be very sensitive to perturbations, Padé approximations with order N>10 should be avoided.

References

[1] Golub, G. H. and C. F. Van Loan, *Matrix Computations*, Johns Hopkins University Press, Baltimore, 1989, pp. 557–558.

pade

See Also

c2d delay2z Discretization of continuous system Changes transfer functions of discrete-time LTI models with delays to rational functions or absorbs FRD delays into the frequency response phase information

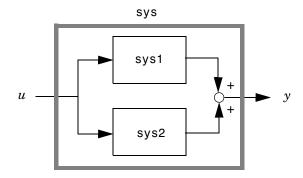
Parallel connection of two LTI models

Syntax

Description

parallel connects two LTI models in parallel. This function accepts any type of LTI model. The two systems must be either both continuous or both discrete with identical sample time. Static gains are neutral and can be specified as regular matrices.

sys = parallel(sys1, sys2) forms the basic parallel connection shown below.



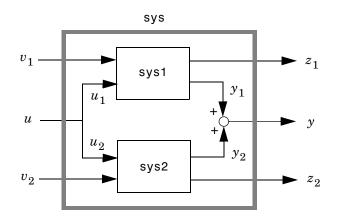
This command is equivalent to the direct addition

$$sys = sys1 + sys2$$

(See Addition and Subtraction for details on LTI system addition.)

parallel

sys = parallel(sys1,sys2,inp1,inp2,out1,out2) forms the more general
parallel connection.



The index vectors inp1 and inp2 specify which inputs u_1 of sys1 and which inputs u_2 of sys2 are connected. Similarly, the index vectors out1 and out2 specify which outputs y_1 of sys1 and which outputs y_2 of sys2 are summed. The resulting model sys has $[v_1 \; ; u \; ; v_2]$ as inputs and $[z_1 \; ; y \; ; z_2]$ as outputs.

Example

See Kalman Filtering for an example.

See Also

append	Append LTI systems
feedback	Feedback connection
series	Series connection

Pole placement design

Syntax

```
K = place(A,B,p)
[K,prec,message] = place(A,B,p)
```

Description

Given the single- or multi-input system

$$\dot{x} = Ax + Bu$$

and a vector p of desired self-conjugate closed-loop pole locations, place computes a gain matrix K such that the state feedback u = -Kx places the closed-loop poles at the locations p. In other words, the eigenvalues of A - BK match the entries of p (up to the ordering).

K = place(A,B,p) computes a feedback gain matrix K that achieves the desired closed-loop pole locations p, assuming all the inputs of the plant are control inputs. The length of p must match the row size of A. place works for multi-input systems and is based on the algorithm from [1]. This algorithm uses the extra degrees of freedom to find a solution that minimizes the sensitivity of the closed-loop poles to perturbations in A or B.

[K,prec,message] = place(A,B,p) also returns prec, an estimate of how closely the eigenvalues of A-BK match the specified locations p (prec measures the number of accurate decimal digits in the actual closed-loop poles). If some nonzero closed-loop pole is more than 10% off from the desired location, message contains a warning message.

You can also use place for estimator gain selection by transposing the A matrix and substituting C' for B.

```
1 = place(A',C',p).'
```

Example

Consider a state-space system (a,b,c,d) with two inputs, three outputs, and three states. You can compute the feedback gain matrix needed to place the closed-loop poles at $p = [1.1 \ 23 \ 5.0]$ by

```
p = [1 \ 1.23 \ 5.0];

K = place(a,b,p)
```

place

Algorithm

place uses the algorithm of [1] which, for multi-input systems, optimizes the choice of eigenvectors for a robust solution. We recommend place rather than acker even for single-input systems.

In high-order problems, some choices of pole locations result in very large gains. The sensitivity problems attached with large gains suggest caution in the use of pole placement techniques. See [2] for results from numerical testing.

References

[1] Kautsky, J. and N.K. Nichols, "Robust Pole Assignment in Linear State Feedback," *Int. J. Control*, 41 (1985), pp. 1129–1155.

[2] Laub, A.J. and M. Wette, *Algorithms and Software for Pole Assignment and Observers*, UCRL-15646 Rev. 1, EE Dept., Univ. of Calif., Santa Barbara, CA, Sept. 1984.

See Also

acker Pole placement using Ackermann's formula

lqr State-feedback LQ regulator design

rlocus Root locus design

Purpose Compute the poles of an LTI system

Syntax p = pole(sys)

Limitations

Description pole computes the poles p of the SISO or MIMO LTI model sys.

Algorithm For state-space models, the poles are the eigenvalues of the A matrix, or the generalized eigenvalues of $A - \lambda E$ in the descriptor case.

For SISO transfer functions or zero-pole-gain models, the poles are simply the denominator roots (see roots).

For MIMO transfer functions (or zero-pole-gain models), the poles are computed as the union of the poles for each SISO entry. If some columns or rows have a common denominator, the roots of this denominator are counted only once.

Multiple poles are numerically sensitive and cannot be computed to high accuracy. A pole λ with multiplicity m typically gives rise to a cluster of computed poles distributed on a circle with center λ and radius of order

 $\rho \approx \epsilon^{1/m}$

where ϵ is the relative machine precision (eps).

See Also Damping and natural frequency of system poles

esort, dsort Sort system poles pzmap Pole-zero map

zero Compute (transmission) zeros

Compute the pole-zero map of an LTI model

Syntax

Description

pzmap(sys) plots the pole-zero map of the continuous- or discrete-time LTI model sys. For SISO systems, pzmap plots the transfer function poles and zeros. For MIMO systems, it plots the system poles and transmission zeros. The poles are plotted as x's and the zeros are plotted as o's.

pzmap(sys1,sys2,...,sysN) plots the pole-zero map of several LTI models on a single figure. The LTI models can have different numbers of inputs and outputs and can be a mix of continuous and discrete systems.

When invoked with left-hand arguments,

$$[p,z] = pzmap(sys)$$

returns the system poles and (transmission) zeros in the column vectors p and z. No plot is drawn on the screen.

You can use the functions sgrid or zgrid to plot lines of constant damping ratio and natural frequency in the s - or z -plane.

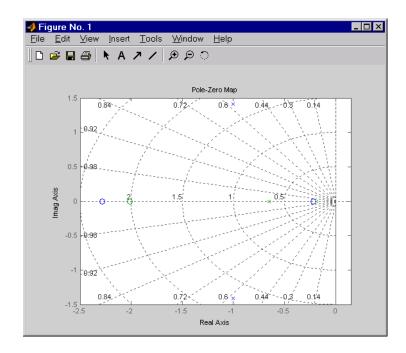
Example

Plot the poles and zeros of the continuous-time system.

$$H(s) = \frac{2s^2 + 5s + 1}{s^2 + 2s + 3}$$

$$H = tf([2 5 1],[1 2 3]); sgrid$$

pzmap(H)



Algorithm

pzmap uses a combination of pole and zero.

See Also

damp	Damping and natural frequency of system poles
esort, dsort	Sort system poles
pole	Compute system poles
rlocus	Root locus
sgrid, zgrid	Plot lines of constant damping and natural frequency
zero	Compute system (transmission) zeros

Compute the pole-zero map of an LTI model and return the plot handle

Syntax

```
h = pzplot(sys1,sys2,...)
h = pzplot(AX,...)
h = pzplot(..., plotoptions)
```

Description

h = pzplot(sys) computes the poles and (transmission) zeros of the LTI model sys and plots them in the complex plane. The poles are plotted as x's and the zeros are plotted as o's. It also returns the plot handle h. You can use this handle to customize the plot with the getoptions and setoptions commands. Type

```
help pzoptions
```

h = pzplot(svs)

for a list of available plot options.

pzplot(sys1, sys2,...) shows the poles and zeros of multiple LTI models sys1,sys2,... on a single plot. You can specify distinctive colors for each model, as in

```
pzplot(sys1,'r',sys2,'y',sys3,'g')
```

 $\mathsf{pzplot}(\mathsf{AX},\ldots)$ plots into the axes with handle AX.

 $\mathsf{pzplot}(\ldots,\,\mathsf{plotoptions})$ plots the poles and zeros with the options specified in $\mathsf{plotoptions}$. Type

```
help pzoptions
```

for more detail.

The function sgrid or zgrid can be used to plot lines of constant damping ratio and natural frequency in the *s*- or *z*-plane.

For arrays sys of LTI models, pzmap plots the poles and zeros of each model in the array on the same diagram.

Example

Use the plot handle to change the color of the plot's title.

```
sys = rss(3,2,2);
h = rlocusplot(sys);
```

```
p = getoptions(h); % Get options for plot.
p.Title.Color = [1,0,0]; % Change title color in options.
setoptions(h,p); % Apply options to plot.
```

See Also

getoptions Get plot options

pzmap Compute the pole-zero map of an LTI model

setoptions Set plot options

Form regulator given state-feedback and estimator gains

Syntax

Description

rsys = reg(sys,K,L) forms a dynamic regulator or compensator rsys given a state-space model sys of the plant, a state-feedback gain matrix K, and an estimator gain matrix L. The gains K and L are typically designed using pole placement or LQG techniques. The function reg handles both continuous- and discrete-time cases.

This syntax assumes that all inputs of sys are controls, and all outputs are measured. The regulator rsys is obtained by connecting the state-feedback law u = -Kx and the state estimator with gain matrix L (see estim). For a plant with equations

$$\dot{x} = Ax + Bu$$

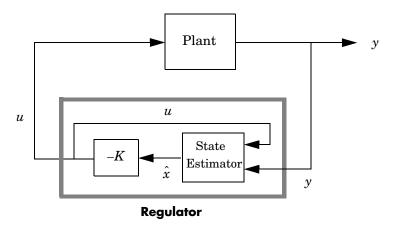
$$v = Cx + Du$$

this yields the regulator

$$\hat{x} = [A - LC - (B - LD)K] \hat{x} + Ly$$

$$u = -K\hat{x}$$

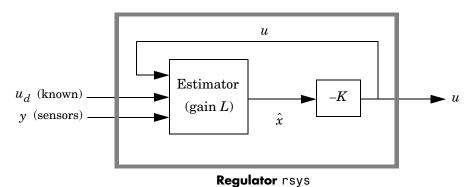
This regulator should be connected to the plant using *positive* feedback.



rsys = reg(sys,K,L,sensors,known,controls) handles more general
regulation problems where:

- \bullet The plant inputs consist of controls u , known inputs u_d , and stochastic inputs w .
- \bullet Only a subset y of the plant outputs is measured.

The index vectors sensors, known, and controls specify y, u_d , and u as subsets of the outputs and inputs of sys. The resulting regulator uses $[u_d; y]$ as inputs to generate the commands u (see figure below).



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Example

Given a continuous-time state-space model

```
sys = ss(A,B,C,D)
```

with seven outputs and four inputs, suppose you have designed:

- A state-feedback controller gain K using inputs 1, 2, and 4 of the plant as control inputs
- A state estimator with gain L using outputs 4, 7, and 1 of the plant as sensors, and input 3 of the plant as an additional known input

You can then connect the controller and estimator and form the complete regulation system by

```
controls = [1,2,4];
sensors = [4,7,1];
known = [3];
regulator = reg(sys,K,L,sensors,known,controls)
```

See Also

estim	Form	state	esti	mator	given	estimator	gain
	1						

kalman Kalman estimator design lqgreg Form LQG regulator

lgr, dlgr State-feedback LQ regulator

place Pole placement

Purpose Change the shape of an LTI array

Description sys = reshape(sys, s1, s2,...,sk) (or, equival

sys = reshape(sys,s1,s2,...,sk) (or, equivalently, sys = reshape(sys,[s1 s2 ... sk])) reshapes the LTI array sys into an s1-by-s2-by...-sk array of LTI models. Equivalently, sys = reshape(sys,[s1 s2 ... sk]) reshapes the LTI array sys into an s1-by-s2-by...-sk array of LTI models. With either syntax, there must be s1*s2*...*sk models in sys to begin with.

Example

```
sys = rss(4,1,1,2,3);
size(sys)

2x3 array of state-space models
Each model has 1 output, 1 input, and 4 states.
sys1 = reshape(sys,6);
size(sys1)

6x1 array of state-space models
Each model has 1 output, 1 input, and 4 states.
```

See Also

ndims Provide the number of dimensions of an LTI array size Provide the lengths of each dimension of an LTI array

rlocus

Purpose

Evans root locus

Syntax

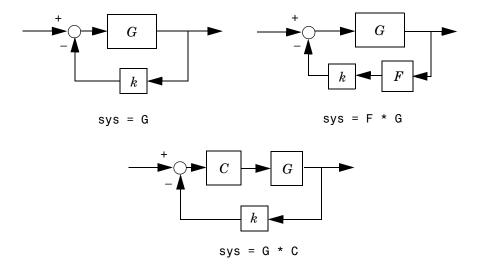
```
rlocus(sys)
rlocus(sys,k)
rlocus(sys1,sys2,...)
[r,k] = rlocus(sys)
```

r = rlocus(sys,k)

Description

rlocus computes the Evans root locus of a SISO open-loop model. The root locus gives the closed-loop pole trajectories as a function of the feedback gain k (assuming negative feedback). Root loci are used to study the effects of varying feedback gains on closed-loop pole locations. In turn, these locations provide indirect information on the time and frequency responses.

rlocus (sys) calculates and plots the root locus of the open-loop SISO model sys. This function can be applied to any of the following *negative* feedback loops by setting sys appropriately.



If sys has transfer function

$$h(s) = \frac{n(s)}{d(s)}$$

the closed-loop poles are the roots of

$$d(s) + k \ n(s) = 0$$

rlocus adaptively selects a set of positive gains k to produce a smooth plot. Alternatively,

uses the user-specified vector k of gains to plot the root locus.

rlocus(sys1,sys2,...) draws the root loci of multiple LTI models sys1, sys2,... on a single plot. You can specify a color, line style, and marker for each model, as in

When invoked with output arguments,

return the vector k of selected gains and the complex root locations r for these gains. The matrix r has length(k) columns and its jth column lists the closed-loop roots for the gain k(j).

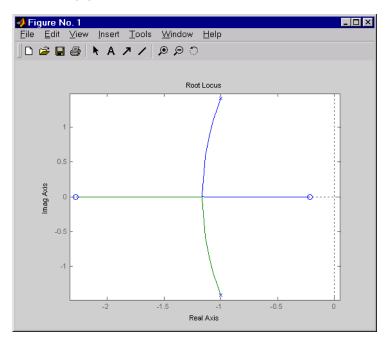
Example

Find and plot the root-locus of the following system.

$$h(s) = \frac{2s^2 + 5s + 1}{s^2 + 2s + 3}$$

$$h = tf([2 5 1],[1 2 3]);$$





You can use the right-click menu for rlocus to add grid lines, zoom in or out, and invoke the Property Editor to customize the plot. Also, click anywhere on the curve to activate a data marker that displays the gain value, pole, damping, overshoot, and frequency at the selected point.

See Also

pole	System poles
pzmap	Pole-zero map

Calculate the root locus and return the plot handle

Syntax

```
h = rlocusplot(sys,k)
h = rlocusplot(sys1,sys2,...)
h = rlcousplot(AX,...)
h = rlocusplot(..., plotoptions)
```

Description

h = rlocusplot(sys) computes and plots the root locus of the single-input, single-output LTI model sys. It also returns the plot handle h. You can use this handle to customize the plot with the getoptions and setoptions commands. Type

```
help pzoptions
```

h = rlocusplot(sys)

for a list of available plot options.

See rlocus for a discussion of the feedback structure and algorithms used to calculate the root locus.

rlocusplot(sys,k) uses a user-specified vector k of gain values.

rlocusplot (sys1, sys2,...) draws the root loci of multiple LTI models sys1, sys2,... on a single plot. You can specify a color, line style, and marker for each model, as in

```
rlocusplot(sys1,'r',sys2,'y:',sys3,'gx')
```

rlocusplot(AX,...) plots into the axes with handle AX.

rlocusplot(..., plotoptions) plots the root locus with the options specified in plotoptions. Type

```
help pzoptions
```

for more details.

Example

Use the plot handle to change the title of the plot.

```
sys = rss(3,2,2);
h = rlocusplot(sys);
p = getoptions(h); % Get options for plot.
```

rlocusplot

p.Title.String = 'My Title'; % Change title in options.
setoptions(h,p); % Apply options to plot.

See Also

getoptions Get plot options

rlocus Calculate root loci of LTI models

setoptions Set plot options

Generate stable random continuous test models

Syntax

```
sys = rss(n)
sys = rss(n,p)
sys = rss(n,p,m)
sys = rss(n,p,m,s1,...,sn)
```

Description

rss(n) produces a stable random n-th order model with one input and one output and returns the model in the state-space object sys.

rss(n,p) produces a random nth order stable model with one input and p outputs, and rss(n,m,p) produces a random n-th order stable model with m inputs and p outputs. The output sys is always a state-space model.

rss(n,p,m,s1,...,sn) produces an s1-by-...-by-sn array of random n-th order stable state-space models with m inputs and p outputs.

Use tf, frd, or zpk to convert the state-space object sys to transfer function, frequency response, or zero-pole-gain form.

Example

Obtain a stable random continuous LTI model with three states, two inputs, and two outputs by typing

sys = rss(3,2,2)			
a =			
	x1	x2	х3
x1	-0.54175	0.09729	0.08304
x2	0.09729	-0.89491	0.58707
x3	0.08304	0.58707	-1.95271
b =			
	u1	u2	
x1	-0.88844	-2.41459	
x2	0	-0.69435	
x3	-0.07162	-1.39139	
0 -			
C =	4	w0	w0
	x1	x2	x3
y1	0.32965	0.14718	0
y2	0.59854	-0.10144	0.02805

Continuous-time system.

See	Δ	so

drss	Generate stable random discrete test models
frd	Convert LTI systems to frequency response form
tf	Convert LTI systems to transfer function form
zpk	Convert LTI systems to zero-pole-gain form

Series connection of two LTI models

Syntax

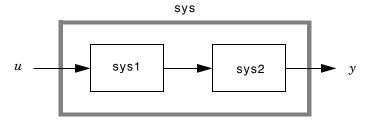
sys = series(sys1,sys2)

sys = series(sys1,sys2,outputs1,inputs2)

Description

series connects two LTI models in series. This function accepts any type of LTI model. The two systems must be either both continuous or both discrete with identical sample time. Static gains are neutral and can be specified as regular matrices.

sys = series(sys1,sys2) forms the basic series connection shown below.

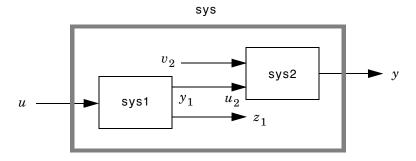


This command is equivalent to the direct multiplication

$$sys = sys2 * sys1$$

See Multiplication for details on multiplication of LTI models.

sys = series(sys1,sys2,outputs1,inputs2) forms the more general series connection.



series

The index vectors outputs 1 and inputs 2 indicate which outputs y_1 of sys1 and which inputs u_2 of sys2 should be connected. The resulting model sys has u as input and y as output.

Example

Consider a state-space system sys1 with five inputs and four outputs and another system sys2 with two inputs and three outputs. Connect the two systems in series by connecting outputs 2 and 4 of sys1 with inputs 1 and 2 of sys2.

```
outputs1 = [2 4];
inputs2 = [1 2];
sys = series(sys1,sys2,outputs1,inputs2)
```

See Also

append Append LTI systems feedback Feedback connection Parallel connection

Set or modify LTI model properties

Syntax

```
set(sys, 'Property', Value)
set(sys, 'Property1', Value1, 'Property2', Value2,...)
set(sys, 'Property')
set(sys)
```

Description

set is used to set or modify the properties of an LTI model (see "LTI Properties" for background on LTI properties). Like its Handle Graphics[®] counterpart, set uses property name/property value pairs to update property values.

set(sys, 'Property', Value) assigns the value Value to the property of the LTI model sys specified by the string 'Property'. This string can be the full property name (for example, 'UserData') or any unambiguous case-insensitive abbreviation (for example, 'user'). The specified property must be compatible with the model type. For example, if sys is a transfer function, Variable is a valid property but StateName is not (see "Model-Specific Properties" for details).

set(sys, 'Property1', Value1, 'Property2', Value2,...) sets multiple property values with a single statement. Each property name/property value pair updates one particular property.

set(sys, 'Property') displays admissible values for the property specified by 'Property'. See "Property Values" below for an overview of legitimate LTI property values.

set(sys) displays all assignable properties of sys and their admissible values.

Example

Consider the SISO state-space model created by

```
sys = ss(1,2,3,4);
```

You can add an input delay of 0.1 second, label the input as torque, reset the D matrix to zero, and store its DC gain in the 'Userdata' property by

```
set(sys,'inputd',0.1,'inputn','torque','d',0,'user',dcgain(sys))
```

Note that set does not require any output argument. Check the result with get by typing

```
get(sys)
       a = 1
        b = 2
        c = 3
        d = 0
        e = []
        Nx = 1
        StateName = {''}
        Ts = 0
        InputDelay = 0.1
        OutputDelay = 0
        ioDelay = 0
        InputName = {'torque'}
        OutputName = {''}
        InputGroup = {0x2 cell}
        OutputGroup = {0x2 cell}
        Notes = \{\}
        UserData = -6
```

Property Values

The following table lists the admissible values for each LTI property. N_u and N_y denotes the number of inputs and outputs of the underlying LTI model. For K-dimensional LTI arrays, let S_1, S_2, \ldots, S_K denote the array dimensions.

Table 1-2: LTI Properties

Property Name	Admissible Property Values
Ts	 0 (zero) for continuous-time systems Sample time in seconds for discrete-time systems -1 or [] for discrete systems with unspecified sample time Note: Resetting the sample time property does not alter the model data. Use c2d, d2c, or d2d for discrete/continuous and discrete/discrete conversions.
ioDelay	Input/Output delays specified with • Nonnegative real numbers for continuous-time models (seconds) • Integers for discrete-time models (number of sample periods) • Scalar when all I/O pairs have the same delay • N_y -by- N_u matrix to specify independent delay times for each I/O pair • Array of size N_y -by- N_u -by- S_1 -byby- S_n to specify different I/O delays for each model in an LTI array.
InputDelay	Input delays specified with • Nonnegative real numbers for continuous-time models (seconds) • Integers for discrete-time models (number of sample periods) • Scalar when $N_u=1$ or system has uniform input delay • Vector of length N_u to specify independent delay times for each input channel • Array of size N_y -by- N_u -by- S_1 -byby- S_n to specify different input delays for each model in an LTI array.

Table 1-2: LTI Properties (Continued)

Property Name	Admissible Property Values
OutputDelay	Output delays specified with
	• Nonnegative real numbers for continuous-time models (seconds)
	• Integers for discrete-time models (number of sample periods)
	• Scalar when $N_y = 1$ or system has uniform output delay
	$ullet$ Vector of length N_y to specify independent delay times for each output channel
	\bullet Array of size N_y -by- N_u -by- S_1 -byby- S_n to specify different output delays for each model in an LTI array.
Notes	String, array of strings, or cell array of strings
UserData	Arbitrary MATLAB variable
InputName	• String for single-input systems, for example, 'thrust'
	• Cell vector of strings for multi-input systems (with as many cells as inputs), for example, {'u'; 'w'} for a two-input system
	 Padded array of strings with as many rows as inputs, for example, ['rudder'; 'aileron']
OutputName	Same as InputName (with "input" replaced by "output")
InputGroup	Cell array. See "Input Groups and Output Groups."
OutputGroup	Same as InputGroup

Table 1-3: State-Space Model Properties

Property Name	Admissible Property Values
StateName	Same as InputName (with Input replaced by State)
a, b, c, d, e	Real- or complex-valued state-space matrices (multidimensional arrays, in the case of LTI arrays) with compatible dimensions for the number of states, inputs, and outputs. See "The Size of LTI Array Data for SS Models."
Nx	• Scalar integer representing the number of states for single LTI models or LTI arrays with the same number of states in each model $ \bullet S_1 - \text{by-} \dots - \text{by-} S_K - \text{dimensional array of integers when all of the models of an LTI array do not have the same number of states } $

Table 1-4: TF Model Properties

Property Name	Admissible Property Values
num, den	• Real- or complex-valued row vectors for the coefficients of the numerator or denominator polynomials in the SISO case. List the coefficients in <i>descending</i> powers of the variable s or z by default, and in <i>ascending</i> powers of $q = z^{-1}$ when the Variable property is set to 'q' or 'z^-1' (see note below).
	• N_y -by- N_u cell arrays of real- or complex-valued row vectors in the MIMO case, for example, {[1 2];[1 0 3]} for a two-output/one-input transfer function
	• N_y -by- N_u -by- S_1 -byby- S_K -dimensional real- or complex-valued cell arrays for MIMO LTI arrays
Variable	 String 's' (default) or 'p' for continuous-time systems String 'z' (default), 'q', or 'z^-1' for discrete-time systems

Table 1-5: ZPK Model Properties

Property Name	Admissible Property Values
z, p	• Vectors of zeros and poles (either real- or complex-valued) in SISO case • N_y -by- N_u cell arrays of vectors (entries are real- or complex valued) in MIMO case, for example, z = {[],[-1 0]} for a model with two inputs and one output • N_y -by- N_u -by- S_1 -byby- S_K -dimensional cell arrays for MIMO LTI arrays
Variable	 String 's' (default) or 'p' for continuous-time systems String 'z' (default), 'q', or 'z^-1' for discrete-time systems

Table 1-6: FRD Model Properties

Property Name	Admissible Property Values
Frequency	Real-valued vector of length N_f -by-1, where N_f is the number of frequencies
Response	$ \bullet \ \ N_y \ \text{-by-} N_u \ \text{-by-} N_f \ \text{-dimensional array of complex data for single LTI models} $ $ \bullet \ \ N_y \ \text{-by-} N_u \ \text{-by-} N_f \ \text{-by-} S_1 \ \text{-by-} \dots \text{-by-} S_K \ \text{-dimensional array for LTI arrays} $
Units	String 'rad/s' (default), or 'Hz'

Remark

For discrete-time transfer functions, the convention used to represent the numerator and denominator depends on the choice of variable (see the tf entry for details). Like tf, the syntax for set changes to remain consistent with the choice of variable. For example, if the Variable property is set to 'z' (the default),

produces the transfer function

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

However, if you change the Variable to 'z^-1' (or 'q') by

the same command

now interprets the row vectors [1 2] and [1 3 4] as the polynomials $1+2z^{-1}$ and $1+3z^{-1}+4z^{-2}$ and produces:

$$\overline{h}(z^{-1}) = \frac{1 + 2z^{-1}}{1 + 3z^{-1} + 4z^{-2}} = zh(z)$$

Note Because the resulting transfer functions are different, make sure to use the convention consistent with your choice of variable.

See Also

get	Access/query LTI model properties
frd	Specify a frequency response data model
SS	Specify a state-space model
tf	Specify a transfer function
zpk	Specify a zero-pole-gain model

setoptions

Purpose

Set plot options for response plots

Syntax

```
setoptions(h, PlotOpts)
setoptions(h, 'Property1', 'value1', ...)
setoptions(h, PlotOpts, 'Property1', 'value1', ...)
```

Description

setoptions(h, PlotOpts) sets preferences for response plot using the plot handle. h is the plot handle, PlotOpts is a plot options handle containing information about plot options.

There are two ways to create a plot options handle:

 Use getoptions, which accepts a plot handle and returns a plot options handle.

```
p = getoptions(h)
```

- Create a default plot options handle using one of the following commands:
 - bodeoptions Bode plots
 - hsvoptions Hankel singular values plots
 - nicholsoptions Nichols plots
 - nyquistoptions Nyquist plots
 - pzoptions Pole/zero plots
 - sigmaoptions Sigma plots
 - timeoptions Time plots (step, initial, impulse, etc.)

For example,

```
p = bodeoptions
```

returns a plot options handle for Bode plots.

setoptions(h, 'Property1', 'value1', ...) assigns values to property pairs instead of using Plot0pts. To find out what properties and values are available, type help <function>options. For example, for Bode plots type

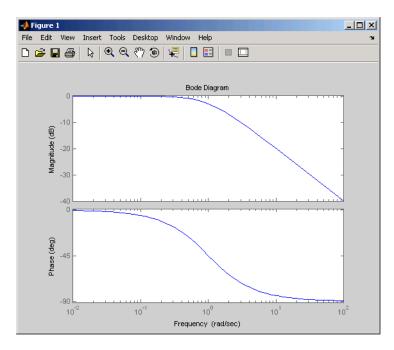
```
help bodeoptions
```

setoptions(h, PlotOpts, 'Property1', 'value1', ...) first assigns plot properties as defined in @PlotOptions, and then overrides any properties governed by the specified property/value pairs.

Examples

To change frequency units, first create a Bode plot.

```
sys=tf(1,[1 1]);
h-bodeplot(sys) % Create a Bode plot with plot handle h.
```

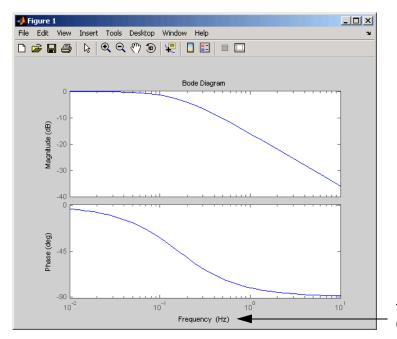


Now, change the frequency units from rad/s to Hz.

```
p=getoptions(h); % Create a plot options handle p.
p.FreqUnits = 'Hz'; % Modify frequency units.
```

setoptions

setoptions(h,p); % Apply plot options to the Bode plot and % render.



The frequency units are now Hz.

To change the frequency units using property/value pairs, use this code.

```
sys=tf(1,[1 1]);
h=bodeplot(sys);
setoptions(h,'FreqUnits','Hz');
```

The result is the same as the first example.

See Also

getoptions

Get a plot options handle

Generate an s-plane grid of constant damping factors and natural frequencies

Syntax

Description

sgrid generates, for pole-zero and root locus plots, a grid of constant damping factors from zero to one in steps of 0.1 and natural frequencies from zero to 10 rad/sec in steps of one rad/sec, and plots the grid over the current axis. If the current axis contains a continuous s-plane root locus diagram or pole-zero map, sgrid draws the grid over the plot.

sgrid(z,wn) plots a grid of constant damping factor and natural frequency lines for the damping factors and natural frequencies in the vectors z and wn, respectively. If the current axis contains a continuous s-plane root locus diagram or pole-zero map, sgrid(z,wn) draws the grid over the plot.

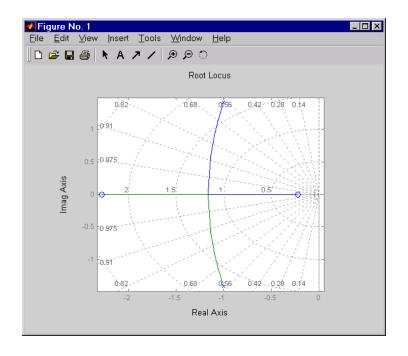
Alternatively, you can select **Grid** from the right-click menu to generate the same s-plane grid.

Example

Plot s-plane grid lines on the root locus for the following system.

$$H(s) = \frac{2s^2 + 5s + 1}{s^2 + 2s + 3}$$

You can do this by typing



See Also

pzmap rlocus zgrid Plot pole-zero map Plot root locus Generate *z*-plane grid lines

Singular values of the frequency response of LTI models

Syntax

```
sigma(sys)
sigma(sys,w)
sigma(sys,w,type)

sigma(sys1,sys2,...,sysN)
sigma(sys1,sys2,...,sysN,w)
sigma(sys1,sys2,...,sysN,w,type)
sigma(sys1,'PlotStyle1',...,sysN,'PlotStyleN')

[sv,w] = sigma(sys)
sv = sigma(sys,w)
```

Description

sigma calculates the singular values of the frequency response of an LTI model. For an FRD model, sys, sigma computes the singular values of sys.Response at the frequencies, sys.frequency. For continuous-time TF, SS, or ZPK models with transfer function H(s), sigma computes the singular values of $H(j\omega)$ as a function of the frequency ω . For discrete-time TF, SS, or ZPK models with transfer function H(z) and sample time T_s , sigma computes the singular values of

$$H(e^{j\omega T_s})$$

for frequencies ω between 0 and the Nyquist frequency ω_N = π/T_s .

The singular values of the frequency response extend the Bode magnitude response for MIMO systems and are useful in robustness analysis. The singular value response of a SISO system is identical to its Bode magnitude response. When invoked without output arguments, sigma produces a singular value plot on the screen.

sigma(sys) plots the singular values of the frequency response of an arbitrary LTI model sys. This model can be continuous or discrete, and SISO or MIMO. The frequency points are chosen automatically based on the system poles and zeros, or from sys.frequency if sys is an FRD.

sigma(sys,w) explicitly specifies the frequency range or frequency points to be used for the plot. To focus on a particular frequency interval [wmin,wmax], set

w = {wmin, wmax}. To use particular frequency points, set w to the corresponding vector of frequencies. Use logspace to generate logarithmically spaced frequency vectors. The frequencies must be specified in rad/sec.

sigma(sys,[],type) or sigma(sys,w,type) plots the following modified singular value responses:

```
type = 1 Singular values of the frequency response H^{-1}, where H is the frequency response of sys.

type = 2 Singular values of the frequency response I + H.

type = 3 Singular values of the frequency response I + H^{-1}.
```

These options are available only for square systems, that is, with the same number of inputs and outputs.

To superimpose the singular value plots of several LTI models on a single figure, use

```
sigma(sys1,sys2,...,sysN)
sigma(sys1,sys2,...,sysN,[],type) % modified SV plot
sigma(sys1,sys2,...,sysN,w) % specify frequency range/grid
```

The models sys1, sys2,..., sysN need not have the same number of inputs and outputs. Each model can be either continuous- or discrete-time. You can also specify a distinctive color, linestyle, and/or marker for each system plot with the syntax

```
sigma(sys1, 'PlotStyle1',...,sysN, 'PlotStyleN')
```

See bode for an example.

When invoked with output arguments,

```
[sv,w] = sigma(sys)
sv = sigma(sys,w)
```

return the singular values sv of the frequency response at the frequencies w. For a system with Nu input and Ny outputs, the array sv has min(Nu, Ny) rows and as many columns as frequency points (length of w). The singular values at the frequency w(k) are given by sv(:,k).

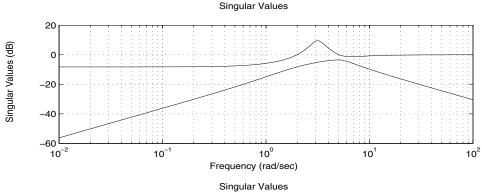
Example

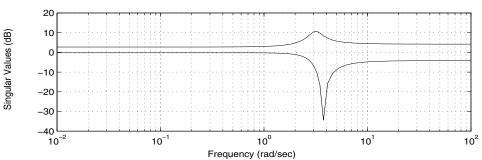
Plot the singular value responses of

$$H(s) = \begin{bmatrix} 0 & \frac{3s}{s^2 + s + 10} \\ \frac{s+1}{s+5} & \frac{2}{s+6} \end{bmatrix}$$

and I + H(s).

You can do this by typing





sigma

Algorithm sigma uses the svd function in MATLAB to compute the singular values of a

complex matrix.

See Also bode Bode plot

evalfr Response at single complex frequency frequesp Frequency response computation

ltiview LTI system viewer

 $\begin{array}{ll} \text{nichols} & \text{Nichols plot} \\ \text{nyquist} & \text{Nyquist plot} \end{array}$

Plot singular values of the frequency response and return the plot handle

Syntax

```
h = sigmaplot(sys,{wmin,wmax})
h = sigmaplot(sys,w)
h = sigmaplot(sys,w,type)
h = sigmaplot(sys,[],type)
h = sigmaplot(AX,...)
h = sigmaplot(..., plotoptions)
```

h = sigmaplot(svs)

Discussion

h = sigmaplot(sys) produces a singular value (SV) plot of the frequency response of the LTI model sys (created with tf, zpk, ss, or frd). It also returns the plot handle h. You can use this handle to customize the plot with the getoptions and setoptions commands. Type

```
help nicholsoptions
```

for a list of available plot options.

The frequency range and number of points are chosen automatically. See bode for details on the notion of frequency in discrete time.

sigmaplot(sys, {wmin, wmax}) draws the SV plot for frequencies ranging between wmin and wmax (in rad/s).

sigmaplot(sys,w) uses the user-supplied vector w of frequencies, in rad/s, at which the frequency response is to be evaluated. See logspace to generate logarithmically spaced frequency vectors.

sigmaplot(sys,w,TYPE) or sigmaplot(sys,[],TYPE) draws the following modified SV plots depending on the value of TYPE:

```
TYPE = 1 \quad --> \quad SV \text{ of } inv(SYS)
TYPE = 2 \quad --> \quad SV \text{ of } I + SYS
TYPE = 3 \quad --> \quad SV \text{ of } I + inv(SYS)
```

sys should be a square system when using this syntax.

sigmaplot(AX,...) plots into the axes with handle AX.

sigmaplot

```
sigmaplot(..., plotoptions) plots the singular values with the options specified in plotoptions. Type
```

```
help sigmaoptions
```

for more details.

Example

Use the plot handle to change the units to Hz.

```
sys = rss(5);
h = sigmaplot(sys);
% Change units to Hz.
setoptions(h,'FreqUnits','Hz');
```

See Also

 $\begin{array}{ll} \text{getoptions} & \quad \text{Get plot options} \\ \text{setoptions} & \quad \text{Set plot options} \end{array}$

sigma Singular value plot of LTI models

Purpose Initialize the SISO Design Tool

Syntax sisotool

sisotool(plant)
sisotool(plant,comp)
sisotool(views)

sisotool(views,plant,comp,sensor,prefilt)

sisotool(views,plant,comp,options)

Description

When invoked without input arguments, sisotool opens a SISO Design GUI for interactive compensator design. This GUI allows you to design a single-input/single-output (SISO) compensator using root locus and Bode diagram techniques.

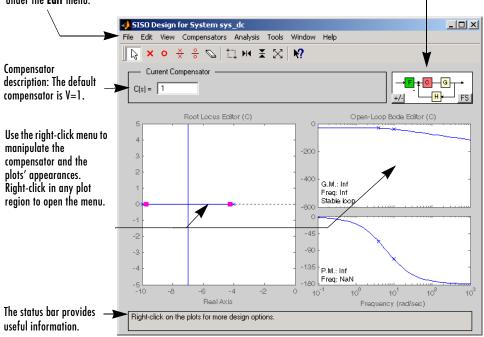
By default, the SISO Design Tool:

- Opens root locus and open-loop Bode diagrams.
- Places the compensator, C, in the forward path in series with the plant, G.
- Assumes the prefilter, F, and the sensor, H, are unity gains. Once you specify
 G and H, they are *fixed* in the feedback structure.

This picture shows the SISO Design Tool.

Use the menu bar to import/export models, and to edit them. Right-click menu functionality is available under the **Edit** menu.

The feedback structure: Click on FS to change the feedback structure. Click on +/- to change the feedback sign.



sisotool(plant) opens the SISO Design Tool, imports plant, and initializes the plant model G to plant. The workspace variable plant can be any SISO LTI model created with either ss, tf, or zpk.

sisotool(plant,comp) initializes the plant model G to plant, the compensator C to comp.

sisotool(plant,comp,sensor,prefilt) initializes the plant ${\bf G}$ to plant, compensator C to comp, sensor H to sensor, and the prefilter F to prefilt. All arguments must be SISO LTI objects.

sisotool(views) or sisotool(views,plant,comp) specifies the initial configuration of the SISO Design Tool. The argument views can be any of the following strings (or combination thereof):

- 'rlocus' Root Locus plot
- 'bode' Bode diagrams of the open-loop response
- 'nichols' Nichols plot
- 'filter' Bode diagrams of the prefilter **F** and the closed-loop response from the command into **F** to the output of the compensator **G** (see the feedback structure figure below)

For example

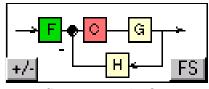
```
sisotool('bode')
```

opens a SISO Design Tool with only the Bode Diagrams on.

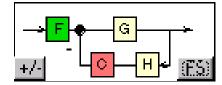
sisotool(plant,comp,options) allows you to override the default compensator location and feedback sign by using an extra input argument options with the following fields:

- options.Location = 'forward' Compensator in the forward loop
- options.Location = 'feedback' Compensator in the feedback loop
- options.Sign = -1 Negative feedback
- options.Sign = 1 Positive feedback

You can design compensators for one of the following two feedback loop configurations.



Compensator in the Forward Path



Compensator in the Feedback Path

The SISO Design Tool Supports Two Feedback Structures.

For more details on the SISO Design Tool, see "Designing Compensators" in the Getting Started documentation for the Control System Toolbox.

See Also

bode Bode response
1tiview Open an LTI Viewer

sisotool

rlocus nichols Root locus

Nichols response

Provide the output/input/array dimensions of LTI models, the model order of TF, SS, and ZPK models, and the number of frequencies of FRD models

Syntax

```
size(sys)
d = size(sys)
Ny = size(sys,1)
Nu = size(sys,2)
Sk = size(sys,2+k)
Ns = size(sys,'order')
Nf = size(sys,'frequency')
```

Description

When invoked without output arguments, size(sys) returns a vector of the number of outputs and inputs for a single LTI model. The lengths of the array dimensions are also included in the response to size when sys is an LTI array. size is the overloaded version of the MATLAB function size for LTI objects.

```
d = size(sys) returns:
```

- The row vector d = [Ny Nu] for a single LTI model sys with Ny outputs and Nu inputs
- The row vector d = [Ny Nu S1 S2 ... Sp] for an S1-by-S2-by-...-by-Sp array of LTI models with Ny outputs and Nu inputs

Ny = size(sys,1) returns the number of outputs of sys.

Nu = size(sys,2) returns the number of inputs of sys.

Sk = size(sys, 2+k) returns the length of the k-th array dimension when sys is an LTI array.

Ns = size(sys,'order') returns the model order of a TF, SS, or ZPK model. This is the same as the number of states for state-space models. When sys is an LTI array, ns is the maximum order of all of the models in the LTI array.

Nf = size(sys, 'frequency') returns the number of frequencies when sys is an FRD. This is the same as the length of sys.frequency.

Example

Consider the random LTI array of state-space models

```
sys = rss(5,3,2,3);
```

Its dimensions are obtained by typing

size

size(sys)

3x1 array of state-space models Each model has 3 outputs, 2 inputs, and 5 states.

See Also

isempty Test if LTI model is empty issiso Test if LTI model is SISO

ndims Number of dimensions of an LTI array

Perform model reduction based on structure

Syntax

msys = sminreal(sys)

Description

msys = sminreal(sys) eliminates the states of the state-space model sys that don't affect the input/output response. All of the states of the resulting state-space model msys are also states of sys and the input/output response of msys is equivalent to that of sys.

sminreal eliminates only structurally non minimal states, i.e., states that can be discarded by looking only at hard zero entries in the A,B, and C matrices. Such structurally nonminimal states arise, for example, when linearizing a Simulink model that includes some unconnected state-space or transfer function blocks.

Remark

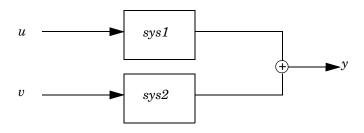
The model resulting from sminreal(sys) is not necessarily minimal, and may have a higher order than one resulting from minreal(sys). However, sminreal(sys) retains the state structure of sys, while, in general, minreal(sys) does not.

Example

Suppose you concatenate two SS models, sys1 and sys2.

$$sys = [sys1, sys2];$$

This operation is depicted in the diagram below.



If you extract the subsystem sys1 from sys, with

sys(1,1)

sminreal

all of the states of sys, including those of sys2 are retained. To eliminate the unobservable states from sys2, while retaining the states of sys1, type

sminreal(sys(1,1))

See Also

minreal

Model reduction by removing

unobservable/uncontrollable states or cancelling

pole/zero pairs

Specify state-space models or convert an LTI model to state space

Syntax

```
sys = ss(a,b,c,d)
sys = ss(a,b,c,d,Ts)
sys = ss(d)
sys = ss(a,b,c,d,ltisys)

sys = ss(a,b,c,d,'Property1',Value1,...,'PropertyN',ValueN)
sys = ss(a,b,c,d,Ts,'Property1',Value1,...,'PropertyN',ValueN)

sys_ss = ss(sys)
sys_ss = ss(sys,'minimal')
```

Description

ss is used to create real- or complex-valued state-space models (SS objects) or to convert transfer function or zero-pole-gain models to state space.

Creation of State-Space Models

sys = ss(a,b,c,d) creates the continuous-time state-space model

$$\dot{x} = Ax + Bu$$

$$y = Cx + Du$$

For a model with Nx states, Ny outputs, and Nu inputs:

- a is an Nx-by-Nx real- or complex-valued matrix.
- ullet b is an Nx-by-Nu real- or complex-valued matrix.
- ullet c is an Ny-by-Nx real- or complex-valued matrix.
- d is an Ny-by-Nu real- or complex-valued matrix.

The output sys is an SS model that stores the model data (see "State-Space Models" on page 2-14). If D=0, you can simply set d to the scalar 0 (zero), regardless of the dimension.

sys = ss(a,b,c,d,Ts) creates the discrete-time model

$$x[n+1] = Ax[n] + Bu[n]$$
$$y[n] = Cx[n] + Du[n]$$

with sample time Ts (in seconds). Set Ts = -1 or Ts = [] to leave the sample time unspecified.

```
sys = ss(d) specifies a static gain matrix D and is equivalent to

sys = ss([],[],[],d)
```

sys = ss(a,b,c,d,ltisys) creates a state-space model with generic LTI properties inherited from the LTI model ltisys (including the sample time). See "Generic Properties" on page 2-26 for an overview of generic LTI properties.

See "Building LTI Arrays" on page 4-12 for information on how to build arrays of state-space models.

Any of the previous syntaxes can be followed by property name/property value pairs.

```
'PropertyName',PropertyValue
```

Each pair specifies a particular LTI property of the model, for example, the input names or some notes on the model history. See the set entry and the example below for details. Note that

```
sys = ss(a,b,c,d,'Property1',Value1,...,'PropertyN',ValueN)
```

is equivalent to the sequence of commands.

```
sys = ss(a,b,c,d)
set(sys,'Property1',Value1,...,'PropertyN',ValueN)
```

Conversion to State Space

sys_ss = ss(sys) converts an arbitrary TF or ZPK model sys to state space. The output sys_ss is an equivalent state-space model (SS object). This operation is known as *state-space realization*.

sys_ss = ss(sys, 'minimal') produces a state-space realization with no
uncontrollable or unobservable states. This is equivalent to sys_ss =
minreal(ss(sys)).

Algorithm

In the case of TF to SS model conversion, ss(sys_tf) returns a modified version of the controllable canonical form. It uses an algorithm similar to

tf2ss, but further rescales the state vector to compress the numerical range in state matrix A and to improve numerics in subsequent computations.

In the case of ZPK to SS conversion, ss(sys_zpk) uses direct form II structures as defined in signal processing texts. See "Discrete-Time Signal Processing" by Oppenheim and Schafer for details.

For example, in the following code, A and sys.a differ by a diagonal state transformation:

See the balance or ssbal documentation for details.

Examples

Example 1

The command

creates a discrete-time model with matrices A, B, C, D and sample time 0.05 second. This model has two states labeled position and velocity, and one input labeled force (the dimensions of A, B, C, D should be consistent with these numbers of states and inputs). Finally, a note is attached with the date of creation of the model.

Example 2

Compute a state-space realization of the transfer function

$$H(s) = \begin{bmatrix} \frac{s+1}{s^3 + 3s^2 + 3s + 2} \\ \frac{s^2 + 3}{s^2 + s + 1} \end{bmatrix}$$

by typing

```
H = [tf([1 1],[1 3 3 2]) ; tf([1 0 3],[1 1 1])];
sys = ss(H);
size(sys)
```

State-space model with 2 outputs, 1 input, and 5 states.

Note that the number of states is equal to the cumulative order of the SISO entries of H(s).

To obtain a minimal realization of H(s), type

```
sys = ss(H,'min');
size(sys)
```

State-space model with 2 outputs, 1 input, and 3 states.

The resulting state-space model order has order three, the minimum number of states needed to represent H(s). This can be seen directly by factoring H(s) as the product of a first order system with a second order one.

$$H(s) = \begin{bmatrix} \frac{1}{s+2} & 0 \\ 0 & 1 \end{bmatrix} \begin{bmatrix} \frac{s+1}{s^2+s+1} \\ \frac{s^2+3}{s^2+s+1} \end{bmatrix}$$

See Also

dss Specify descriptor state-space models.

frd Specify FRD models or convert to an FRD.
get Get properties of LTI models.
set Set properties of LTI models.

ssdata	Retrieve the A, B, C, D matrices of state-space model.
tf	Specify transfer functions or convert to TF.
zpk	Specify zero-pole-gain models or convert to ZPK.

State coordinate transformation for state-space models

Syntax

$$sysT = ss2ss(sys,T)$$

Description

Given a state-space model sys with equations

$$\dot{x} = Ax + Bu$$

$$v = Cx + Du$$

(or their discrete-time counterpart), ss2ss performs the similarity transformation $\bar{x} = Tx$ on the state vector x and produces the equivalent state-space model sysT with equations.

$$\dot{\bar{x}} = TAT^{-1}\bar{x} + TBu$$
$$y = CT^{-1}\bar{x} + Du$$

sysT = ss2ss(sys,T) returns the transformed state-space model sysT given sys and the state coordinate transformation T. The model sys must be in state-space form and the matrix T must be invertible. ss2ss is applicable to both continuous- and discrete-time models.

Example

Perform a similarity transform to improve the conditioning of the $A\,$ matrix.

See ssbal for a more direct approach.

See Also

balreal Grammian-based I/O balancing canon Canonical state-space realizations

ssbal Balancing of state-space models using diagonal

similarity transformations

Balance state-space models using a diagonal similarity transformation

Syntax

Description

Given a state-space model sys with matrices (A, B, C, D),

$$[sysb,T] = ssbal(sys)$$

computes a diagonal similarity transformation T and a scalar α such that

$$\begin{bmatrix} TAT^{-1} & TB/\alpha \\ \alpha CT^{-1} & 0 \end{bmatrix}$$

has approximately equal row and column norms. ssbal returns the balanced model sysb with matrices

$$(TAT^{-1}, TB/\alpha, \alpha CT^{-1}, D)$$

and the state transformation $\bar{x} = Tx$ where \bar{x} is the new state.

[sysb,T] = ssbal(sys,condT) specifies an upper bound condT on the condition number of T. Since balancing with ill-conditioned T can inadvertently magnify rounding errors, condT gives control over the worst-case roundoff amplification factor. The default value is condT=Inf.

ssbal returns an error if the state-space model sys has varying state dimensions.

Example

Consider the continuous-time state-space model with the following data.

$$A = \begin{bmatrix} 1 & 10^4 & 10^2 \\ 0 & 10^2 & 10^5 \\ 10 & 1 & 0 \end{bmatrix}, \qquad B = \begin{bmatrix} 1 \\ 1 \\ 1 \end{bmatrix}, \qquad C = \begin{bmatrix} 0.1 & 10 & 100 \end{bmatrix}$$

ssbal

Balance this model with ssbal by typing

Continuous-time system.

Direct inspection shows that the range of numerical values has been compressed by a factor 100 and that the B and C matrices now have nearly equal norms.

Algorithm

ssbal uses the MATLAB function balance to compute \emph{T} and $\alpha\,.$

See Also

balreal	Grammian-based I/O balancing
ss2ss	State coordinate transformation

Quick access to state-space model data

Syntax

Description

[a,b,c,d] = ssdata(sys) extracts the matrix (or multidimensional array) data (A, B, C, D) from the state-space model (LTI array) sys. If sys is a transfer function or zero-pole-gain model (LTI array), it is first converted to state space. See Table 11-16, "State-Space Model Properties," on page 11-195 for more information on the format of state-space model data.

[a,b,c,d,Ts] = ssdata(sys) also returns the sample time Ts.

You can access the remaining LTI properties of sys with get or by direct referencing, for example,

```
sys.statename
```

For arrays of state-space models with variable numbers of states, use the syntax

```
[a,b,c,d] = ssdata(sys,'cell')
```

to extract the state-space matrices of each model as separate cells in the cell arrays a, b, c, and d.

See Also

dssdata	Quick access to descriptor state-space data
get	Get properties of LTI models
set	Set model properties
SS	Specify state-space models
tfdata	Quick access to transfer function data
zpkdata	Quick access to zero-pole-gain data

stabsep

Purpose

Stable/unstable decomposition of LTI models

Syntax

Description

stapsep decomposes the LTI model into its stable and unstable parts

$$G = GS + GNS$$

where GS contains all stable modes that can be separated from the unstable modes in a numerically stable way, and GNS contains the remaining modes. GNS is always strictly proper.

Use the optional input CONDMAX to control the condition number of the decoupling state transformation. Increasing CONDMAX helps separate close by stable and unstable modes at the expense of accuracy (see BDSCHUR for more details). By default CONDMAX=1e8.

[G1,G2] = STABSEP(G,CONDMAX,MODE,TOL) performs more general stable/unstable decompositions such that G1 includes all separable eigenvalues lying in one the following regions:

Table 1-7:

Mode	Continuous Time	Discrete Time
1	Re(s)<-TOL*max(1, Im(s))	1 z < 1-T0L
2	Re(s) > TOL*max(1, Im(s))	2 z > 1+T0L

The default values are MODE=1 and TOL=0.

See Also

modsep

Region-based modal decomposition

Build an LTI array by stacking LTI models or LTI arrays along array dimensions of an LTI array

Syntax

sys = stack(arraydim,sys1,sys2,...)

Description

sys = stack(arraydim,sys1,sys2,...) produces an array of LTI models sys by stacking (concatenating) the LTI models (or LTI arrays) sys1,sys2,... along the array dimension arraydim. All models must have the same number of inputs and outputs (the same I/O dimensions), but the number of states can vary. The I/O dimensions are not counted in the array dimensions. See "Dimensions, Size, and Shape of an LTI Array" and "Building LTI Arrays Using the stack Function" for more information.

For arrays of state-space models with variable order, you cannot use the dot operator (e.g., sys.a) to access arrays. Use the syntax

```
[a,b,c,d] = ssdata(sys,'cell')
```

to extract the state-space matrices of each model as separate cells in the cell arrays a, b, c, and d.

Example

If sys1 and sys2 are two LTI models:

- \bullet stack(1,sys1,sys2) produces a 2-by-1 LTI array.
- stack(2, sys1, sys2) produces a 1-by-2 LTI array.
- stack(3, sys1, sys2) produces a 1-by-1-by-2 LTI array.

Step response of LTI systems

Syntax

```
step(sys)
step(sys,t)

step(sys1,sys2,...,sysN)
step(sys1,sys2,...,sysN,t)
step(sys1,'PlotStyle1',...,sysN,'PlotStyleN')

[y,t,x] = step(sys)
```

Description

step calculates the unit step response of a linear system. Zero initial state is assumed in the state-space case. When invoked with no output arguments, this function plots the step response on the screen.

step(sys) plots the step response of an arbitrary LTI model sys. This model can be continuous or discrete, and SISO or MIMO. The step response of multi-input systems is the collection of step responses for each input channel. The duration of simulation is determined automatically based on the system poles and zeros.

step(sys,t) sets the simulation horizon explicitly. You can specify either a
final time t = Tfinal (in seconds), or a vector of evenly spaced time samples
of the form

```
t = 0:dt:Tfinal
```

For discrete systems, the spacing dt should match the sample period. For continuous systems, dt becomes the sample time of the discretized simulation model (see "Algorithm"), so make sure to choose dt small enough to capture transient phenomena.

To plot the step responses of several LTI models sys1,..., sysN on a single figure, use

```
step(sys1,sys2,...,sysN)
step(sys1,sys2,...,sysN,t)
```

All systems must have the same number of inputs and outputs but may otherwise be a mix of continuous- and discrete-time systems. This syntax is useful to compare the step responses of multiple systems.

You can also specify a distinctive color, linestyle, and/or marker for each system. For example,

```
step(sys1, 'y:', sys2, 'g--')
```

plots the step response of sys1 with a dotted yellow line and the step response of sys2 with a green dashed line.

When invoked with output arguments,

```
[y,t] = step(sys)
[y,t,x] = step(sys) % for state-space models only
y = step(sys,t)
```

return the output response y, the time vector t used for simulation, and the state trajectories x (for state-space models only). No plot is drawn on the screen. For single-input systems, y has as many rows as time samples (length of t), and as many columns as outputs. In the multi-input case, the step responses of each input channel are stacked up along the third dimension of y. The dimensions of y are then

```
(length of t) \times (number of outputs) \times (number of inputs)
```

and y(:,:,j) gives the response to a unit step command injected in the jth input channel. Similarly, the dimensions of x are

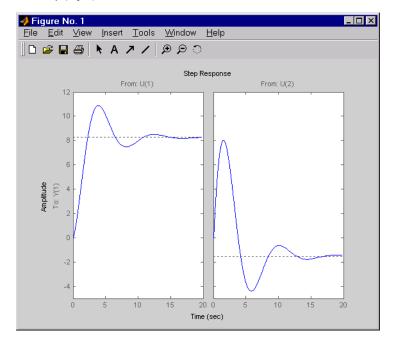
$$(length \ of \ t) \times (number \ of \ states) \times (number \ of \ inputs)$$

Example

Plot the step response of the following second-order state-space model.

$$\begin{bmatrix} \dot{x}_1 \\ \dot{x}_2 \end{bmatrix} = \begin{bmatrix} -0.5572 & -0.7814 \\ 0.7814 & 0 \end{bmatrix} \begin{bmatrix} x_1 \\ x_2 \end{bmatrix} + \begin{bmatrix} 1 & -1 \\ 0 & 2 \end{bmatrix} \begin{bmatrix} u_1 \\ u_2 \end{bmatrix}$$
$$y = \begin{bmatrix} 1.9691 & 6.4493 \end{bmatrix} \begin{bmatrix} x_1 \\ x_2 \end{bmatrix}$$

```
a = [-0.5572  -0.7814;0.7814 0];
b = [1 -1;0 2];
c = [1.9691  6.4493];
sys = ss(a,b,c,0);
step(sys)
```



The left plot shows the step response of the first input channel, and the right plot shows the step response of the second input channel.

Algorithm

Continuous-time models are converted to state space and discretized using zero-order hold on the inputs. The sampling period is chosen automatically based on the system dynamics, except when a time vector t = 0:dt:Tf is supplied (dt is then used as sampling period).

See Also

ımpulse	Impulse response
initial	Free response to initial condition
lsim	Simulate response to arbitrary inputs
ltiview	LTI system viewer

Plot the step response of LTI systems and return the plot handle

Syntax

```
h = stepplot(sys)
```

h = stepplot(sys,Tfinal)

h = stepplot(sys,t)

h = stepplot(sys1, sys2, ..., t)

h = stepplot(AX,...)

h = stepplot(..., plotoptions)

Description

h = stepplot(sys) plots the step response of the LTI model sys (created with either tf, zpk, or ss). It also returns the plot handle h. You can use this handle to customize the plot with the getoptions and setoptions commands. Type

```
help timeoptions
```

for a list of available plot options.

For multiinput models, independent step commands are applied to each input channel. The time range and number of points are chosen automatically.

stepplot(sys,Tfinal) simulates the step response from t=0 to the final time t=Tfinal. For discrete-time models with unspecified sampling time, Tfinal is interpreted as the number of samples.

stepplot(sys,t) uses the user-supplied time vector t for simulation. For discrete-time models, t should be of the form Ti:Ts:Tf, where Ts is the sample time. For continuous-time models, t should be of the form Ti:dt:Tf, where dt becomes the sample time for the discrete approximation to the continuous system. The step input is always assumed to start at t=0 (regardless of Ti).

stepplot(sys1,sys2,...,t) plots the step responses of multiple LTI models sys1,sys2,... on a single plot. The time vector t is optional. You can also specify a color, line style, and marker for each system, as in

```
stepplot(sys1, 'r', sys2, 'y--', sys3, 'gx')
```

stepplot(AX,...) plots into the axes with handle AX.

stepplot(..., plotoptions) plots the step response with the options specified in plotoptions. Type

help timeoptions

stepplot

for more details.

Example

Use the plot handle to normalize the responses on a step plot.

```
sys = rss(3);
h = stepplot(sys,[1,1,1]);
% Normalize responses.
setoptions(h,'Normalize','on');
```

See Also

getoptions Get plot options setoptions Set plot options

step Plot step responses of LTI systems

Specify transfer functions or convert LTI model to transfer function form

Syntax

```
sys = tf(num,den)
sys = tf(num,den,Ts)
sys = tf(M)
sys = tf(num,den,ltisys)

sys = tf(num,den,'Property1',Value1,...,'PropertyN',ValueN)
sys = tf(num,den,Ts,'Property1',Value1,...,'PropertyN',ValueN)

sys = tf('s')
sys = tf('z')

tfsys = tf(sys)
tfsys = tf(sys,'inv') % for state-space sys only
```

Description

tf is used to create real- or complex-valued transfer function models (TF objects) or to convert state-space or zero-pole-gain models to transfer function form.

Creation of Transfer Functions

sys = tf(num,den) creates a continuous-time transfer function with numerator(s) and denominator(s) specified by num and den. The output sys is a TF object storing the transfer function data (see "Transfer Function Models" on page 2-8).

In the SISO case, num and den are the real- or complex-valued row vectors of numerator and denominator coefficients ordered in *descending* powers of s. These two vectors need not have equal length and the transfer function need not be proper. For example, $h = tf([1\ 0],1)$ specifies the pure derivative h(s) = s.

To create MIMO transfer functions, specify the numerator and denominator of each SISO entry. In this case:

• num and den are cell arrays of row vectors with as many rows as outputs and as many columns as inputs.

• The row vectors num{i,j} and den{i,j} specify the numerator and denominator of the transfer function from input j to output i (with the SISO convention).

If all SISO entries of a MIMO transfer function have the same denominator, you can set den to the row vector representation of this common denominator. See "Examples" for more details.

sys = tf(num,den,Ts) creates a discrete-time transfer function with sample time Ts (in seconds). Set Ts = -1 or Ts = [] to leave the sample time unspecified. The input arguments num and den are as in the continuous-time case and must list the numerator and denominator coefficients in *descending* powers of z.

```
sys = tf(M) creates a static gain M (scalar or matrix).
```

sys = tf(num,den,ltisys) creates a transfer function with generic LTI properties inherited from the LTI model ltisys (including the sample time). See "Generic Properties" on page 2-26 for an overview of generic LTI properties.

There are several ways to create LTI arrays of transfer functions. To create arrays of SISO or MIMO TF models, either specify the numerator and denominator of each SISO entry using multidimensional cell arrays, or use a for loop to successively assign each TF model in the array. See "Building LTI Arrays" on page 4-12 for more information.

Any of the previous syntaxes can be followed by property name/property value pairs

```
'Property', Value
```

Each pair specifies a particular LTI property of the model, for example, the input names or the transfer function variable. See set entry and the example below for details. Note that

```
sys = tf(num,den,'Property1',Value1,...,'PropertyN',ValueN)
is a shortcut for
sys = tf(num,den)
set(sys,'Property1',Value1,...,'PropertyN',ValueN)
```

Transfer Functions as Rational Expressions in s or z

You can also use real- or complex-valued rational expressions to create a TF model. To do so, first type either:

- s = tf('s') to specify a TF model using a rational function in the Laplace variable, s.
- z = tf('z',Ts) to specify a TF model with sample time Ts using a rational function in the discrete-time variable, z.

Once you specify either of these variables, you can specify TF models directly as rational expressions in the variable s or z by entering your transfer function as a rational expression in either s or z.

Conversion to Transfer Function

tfsys = tf(sys) converts an arbitrary SS or ZPK LTI model sys to transfer function form. The output tfsys (TF object) is the transfer function of sys. By default, tf uses zero to compute the numerators when converting a state-space model to transfer function form. Alternatively,

uses inversion formulas for state-space models to derive the numerators. This algorithm is faster but less accurate for high-order models with low gain at s=0.

Examples Example 1

Create the two-output/one-input transfer function

$$H(p) = \begin{bmatrix} \frac{p+1}{p^2 + 2p + 2} \\ \frac{1}{p} \end{bmatrix}$$

with input current and outputs torque and ang velocity.

To do this, type

$$num = \{[1 \ 1] \ ; \ 1\}$$

 $den = \{[1 \ 2 \ 2] \ ; \ [1 \ 0]\}$

Note how setting the 'variable' property to 'p' causes the result to be displayed as a transfer function of the variable p.

Example 2

To use a rational expression to create a SISO TF model, type

$$s = tf('s');$$

 $H = s/(s^2 + 2*s +10);$

This produces the same transfer function as

$$h = tf([1\ 0],[1\ 2\ 10]);$$

Example 3

Specify the discrete MIMO transfer function

$$H(z) = \begin{bmatrix} \frac{1}{z+0.3} & \frac{z}{z+0.3} \\ \frac{-z+2}{z+0.3} & \frac{3}{z+0.3} \end{bmatrix}$$

with common denominator d(z) = z + 0.3 and sample time of 0.2 seconds.

```
nums = {1 [1 0];[-1 2] 3}
Ts = 0.2
H = tf(nums,[1 0.3],Ts) % Note: row vector for common den. d(z)
```

Example 4

Compute the transfer function of the state-space model with the following data.

$$A = \begin{bmatrix} -2 & -1 \\ 1 & -2 \end{bmatrix}, \qquad B = \begin{bmatrix} 1 & 1 \\ 2 & -1 \end{bmatrix}, \qquad C = \begin{bmatrix} 1 & 0 \end{bmatrix}, \qquad D = \begin{bmatrix} 0 & 1 \end{bmatrix}$$

To do this, type

$$sys = ss([-2 -1;1 -2],[1 1;2 -1],[1 0],[0 1])$$

 $tf(sys)$

Transfer function from input 1 to output:

Transfer function from input 2 to output:

Example 5

You can use a for loop to specify a 10-by-1 array of SISO TF models.

```
s = tf('s')
H = tf(zeros(1,1,10));
  for k=1:10,
      H(:,:,k) = k/(s^2+s+k);
end
```

The first statement pre-allocates the TF array and fills it with zero transfer functions.

Discrete-Time Conventions

The control and digital signal processing (DSP) communities tend to use different conventions to specify discrete transfer functions. Most control engineers use the z variable and order the numerator and denominator terms in descending powers of z, for example,

$$h(z) = \frac{z^2}{z^2 + 2z + 3}$$

The polynomials z^2 and $z^2 + 2z + 3$ are then specified by the row vectors [1 0 0] and [1 2 3], respectively. By contrast, DSP engineers prefer to write this transfer function as

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

and specify its numerator as 1 (instead of $[1\ 0\ 0]$) and its denominator as $[1\ 2\ 3]$.

tf switches convention based on your choice of variable (value of the 'Variable' property).

Variable	Convention
'z' (default)	Use the row vector [ak a1 a0] to specify the polynomial $a_k z^k + + a_1 z + a_0$ (coefficients ordered in descending powers of z).
'z^-1', 'q'	Use the row vector [b0 b1 bk] to specify the polynomial $b_0 + b_1 z^{-1} + + b_k z^{-k}$ (coefficients in ascending powers of z^{-1} or q).

For example,

$$g = tf([1 1],[1 2 3],0.1)$$

specifies the discrete transfer function

$$g(z) = \frac{z+1}{z^2 + 2z + 3}$$

because z is the default variable. In contrast,

$$h = tf([1 1],[1 2 3],0.1,'variable','z^-1')$$

uses the DSP convention and creates

$$h(z^{-1}) = \frac{1+z^{-1}}{1+2z^{-1}+3z^{-2}} = zg(z)$$

See also filt for direct specification of discrete transfer functions using the DSP convention.

Note that tf stores data so that the numerator and denominator lengths are made equal. Specifically, tf stores the values

```
num = [0 \ 1 \ 1]; den = [1 \ 2 \ 3]
```

for g (the numerator is padded with zeros on the left) and the values

$$num = [1 \ 1 \ 0]; den = [1 \ 2 \ 3]$$

for h (the numerator is padded with zeros on the right).

Algorithm

tf uses the MATLAB function poly to convert zero-pole-gain models, and the functions zero and pole to convert state-space models.

See Also

filt	Specify discrete transfer functions in DSP format
frd	Specify a frequency response data model
get	Get properties of LTI models
set	Set properties of LTI models
SS	Specify state-space models or convert to state space
tfdata	Retrieve transfer function data
zpk	Specify zero-pole-gain models or convert to ZPK

Quick access to transfer function data

Syntax

```
[num,den] = tfdata(sys)
[num,den] = tfdata(sys,'v')
[num,den,Ts] = tfdata(sys)
```

Description

[num,den] = tfdata(sys) returns the numerator(s) and denominator(s) of the transfer function for the TF, SS or ZPK model (or LTI array of TF, SS or ZPK models) sys. For single LTI models, the outputs num and den of tfdata are cell arrays with the following characteristics:

- num and den have as many rows as outputs and as many columns as inputs.
- The (i,j) entries num{i,j} and den{i,j} are row vectors specifying the numerator and denominator coefficients of the transfer function from input j to output i. These coefficients are ordered in *descending* powers of s or z.

For arrays sys of LTI models, num and den are multidimensional cell arrays with the same sizes as sys.

If sys is a state-space or zero-pole-gain model, it is first converted to transfer function form using tf. See Table 11-15, "LTI Properties," on page 11-194 for more information on the format of transfer function model data.

For SISO transfer functions, the syntax

```
[num.den] = tfdata(svs.'v')
```

forces tfdata to return the numerator and denominator directly as row vectors rather than as cell arrays (see example below).

```
[num,den,Ts] = tfdata(sys) also returns the sample time Ts.
```

You can access the remaining LTI properties of sys with get or by direct referencing, for example,

```
sys.Ts
sys.variable
```

Example

Given the SISO transfer function

```
h = tf([1 1],[1 2 5])
```

you can extract the numerator and denominator coefficients by typing

This syntax returns two row vectors.

If you turn h into a MIMO transfer function by typing

$$H = [h ; tf(1,[1 1])]$$

the command

now returns two cell arrays with the numerator/denominator data for each SISO entry. Use celldisp to visualize this data. Type

```
celldisp(num)
```

and MATLAB returns the numerator vectors of the entries of H.

Similarly, for the denominators, type

See Also

 $\begin{array}{ll} \text{get} & \text{Get properties of LTI models} \\ \text{ssdata} & \text{Quick access to state-space data} \end{array}$

tfdata

tf zpkdata Specify transfer functions Quick access to zero-pole-gain data

Return the total combined I/O delays for an LTI model

Syntax

td = totaldelay(sys)

Description

td = totaldelay(sys) returns the total combined I/O delays for an LTI model sys. The matrix td combines contributions from the InputDelay, OutputDelay, and ioDelay properties, (see set on page 11-192 or type ltiprops for details on these properties).

Delays are expressed in seconds for continuous-time models, and as integer multiples of the sample period for discrete-time models. To obtain the delay times in seconds, multiply td by the sample time sys.Ts.

Example

The resulting I/O map is

$$e^{-2s} \times \frac{1}{s} e^{-1.5s} = e^{-3.5s} \frac{1}{s}$$

This is equivalent to assigning an I/O delay of 3.5 seconds to the original model sys.

See Also

delay2z Change transfer functions of discrete-time LTI models

with delays to rational functions or absorbs FRD delays

into the frequency response phase information

hasdelay True for LTI models with delays

Purpose Transm

Transmission zeros of LTI models

Syntax

Description

zero computes the zeros of SISO systems and the transmission zeros of MIMO systems. For a MIMO system with matrices (A,B,C,D), the transmission zeros are the complex values λ for which the normal rank of

$$\begin{bmatrix} A - \lambda I & B \\ C & D \end{bmatrix}$$

drops.

z = zero(sys) returns the (transmission) zeros of the LTI model sys as a column vector.

[z,gain] = zero(sys) also returns the gain (in the zero-pole-gain sense) if sys is a SISO system.

Algorithm

zero is based on SLICOT routine AB08NX. Also use LAPACK routines DGEEV and DGEGV (and their complex counterparts) for eigenvalue computation.

The transmission zeros are computed using the algorithm in [1].

References

[1] Emami-Naeini, A. and P. Van Dooren, "Computation of Zeros of Linear Multivariable Systems," *Automatica*, 18 (1982), pp. 415–430.

See Also

pole Compute the poles of an LTI model pzmap Compute the pole-zero map

Generate a z-plane grid of constant damping factors and natural frequencies

Syntax

zgrid
zgrid(z,wn)

Description

zgrid generates, for root locus and pole-zero maps, a grid of constant damping factors from zero to one in steps of 0.1 and natural frequencies from zero to π in steps of $\pi/10$, and plots the grid over the current axis. If the current axis contains a discrete z-plane root locus diagram or pole-zero map, zgrid draws the grid over the plot without altering the current axis limits.

zgrid(z,wn) plots a grid of constant damping factor and natural frequency lines for the damping factors and normalized natural frequencies in the vectors z and wn, respectively. If the current axis contains a discrete z-plane root locus diagram or pole-zero map, zgrid(z,wn) draws the grid over the plot. The frequency lines for unnormalized (true) frequencies can be plotted using

where Ts is the sample time.

zgrid([],[]) draws the unit circle.

Alternatively, you can select **Grid** from the right-click menu to generate the same z-plane grid.

Example

Plot z-plane grid lines on the root locus for the system

$$H(z) = \frac{2z^2 - 3.4z + 1.5}{z^2 - 1.6z + 0.8}$$

by typing

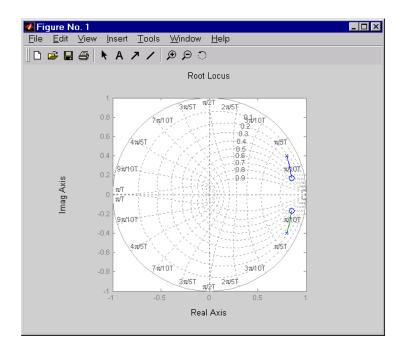
$$H = tf([2 -3.4 1.5],[1 -1.6 0.8],-1)$$

Sampling time: unspecified

zgrid

To see the z-plane grid on the root locus plot, type

```
rlocus(H)
zgrid
axis('square')
```



See Also

pzmap
rlocus
sgrid

Plot pole-zero map of LTI systems Plot root locus Generate s-plane grid lines

Purpose

Specify zero-pole-gain models or convert LTI model to zero-pole-gain form

Syntax

```
sys = zpk(z,p,k)
sys = zpk(z,p,k,Ts)
sys = zpk(M)
sys = zpk(z,p,k,ltisys)

sys = zpk(z,p,k,'Property1',Value1,...,'PropertyN',ValueN)
sys = zpk(z,p,k,Ts,'Property1',Value1,...,'PropertyN',ValueN)

sys = zpk('s')
sys = zpk('s')
sys = zpk(sys)
zsys = zpk(sys,'inv') % for state-space sys only
```

Description

zpk is used to create zero-pole-gain models (ZPK objects) or to convert TF or SS models to zero-pole-gain form.

Creation of Zero-Pole-Gain Models

sys = zpk(z,p,k) creates a continuous-time zero-pole-gain model with zeros z, poles p, and gain(s) k. The output sys is a ZPK object storing the model data (see "LTI Objects" on page 2-3).

In the SISO case, z and p are the vectors of real- or complex-valued zeros and poles, and k is the real- or complex-valued scalar gain.

$$h(s) = k \frac{(s-z(1))(s-z(2))...(s-z(m))}{(s-p(1))(s-p(2))...(s-p(n))}$$

Set z or p to [] for systems without zeros or poles. These two vectors need not have equal length and the model need not be proper (that is, have an excess of poles).

You can also use rational expressions to create a ZPK model. To do so, use either:

• s = zpk('s') to specify a ZPK model from a rational transfer function of the Laplace variable, s.

• z = zpk('z',Ts) to specify a ZPK model with sample time Ts from a rational transfer function of the discrete-time variable, z.

Once you specify either of these variables, you can specify ZPK models directly as real- or complex-valued rational expressions in the variable s or z.

To create a MIMO zero-pole-gain model, specify the zeros, poles, and gain of each SISO entry of this model. In this case:

- z and p are cell arrays of vectors with as many rows as outputs and as many columns as inputs, and k is a matrix with as many rows as outputs and as many columns as inputs.
- The vectors $z\{i,j\}$ and $p\{i,j\}$ specify the zeros and poles of the transfer function from input j to output i.
- k(i,j) specifies the (scalar) gain of the transfer function from input j to output i.

See below for a MIMO example.

sys = zpk(z,p,k,Ts) creates a discrete-time zero-pole-gain model with sample time Ts (in seconds). Set Ts = -1 or Ts = [] to leave the sample time unspecified. The input arguments z, p, k are as in the continuous-time case.

sys = zpk(M) specifies a static gain M.

sys = zpk(z,p,k,ltisys) creates a zero-pole-gain model with generic LTI properties inherited from the LTI model ltisys (including the sample time). See "Generic Properties" on page 2-26 for an overview of generic LTI properties.

To create an array of ZPK models, use a for loop, or use multidimensional cell arrays for z and p, and a multidimensional array for k.

Any of the previous syntaxes can be followed by property name/property value pairs.

'PropertyName', PropertyValue

Each pair specifies a particular LTI property of the model, for example, the input names or the input delay time. See set entry and the example below for details. Note that

```
sys = zpk(z,p,k,'Property1',Value1,...,'PropertyN',ValueN)
```

is a shortcut for the following sequence of commands.

```
sys = zpk(z,p,k)
set(sys, 'Property1', Value1,..., 'PropertyN', ValueN)
```

Zero-Pole-Gain Models as Rational Expressions in s or z

You can also use rational expressions to create a ZPK model. To do so, first type either:

- s = zpk('s') to specify a ZPK model using a rational function in the Laplace variable, s.
- z = zpk('z',Ts) to specify a ZPK model with sample time Ts using a rational function in the discrete-time variable, z.

Once you specify either of these variables, you can specify ZPK models directly as rational expressions in the variable s or z by entering your transfer function as a rational expression in either s or z.

Conversion to Zero-Pole-Gain Form

zsys = zpk(sys) converts an arbitrary LTI model sys to zero-pole-gain form. The output zsys is a ZPK object. By default, zpk uses zero to compute the zeros when converting from state-space to zero-pole-gain. Alternatively,

```
zsvs = zpk(svs,'inv')
```

uses inversion formulas for state-space models to compute the zeros. This algorithm is faster but less accurate for high-order models with low gain at $s\,=\,0$.

Variable Selection

As for transfer functions, you can specify which variable to use in the display of zero-pole-gain models. Available choices include s (default) and p for continuous-time models, and z (default), z^{-1} , or $q=z^{-1}$ for discrete-time models. Reassign the 'Variable' property to override the defaults. Changing the variable affects only the display of zero-pole-gain models.

Example Example 1

Specify the following zero-pole-gain model.

$$H(z) = \begin{vmatrix} \frac{1}{z - 0.3} \\ \frac{2(z + 0.5)}{(z - 0.1 + j)(z - 0.1 - j)} \end{vmatrix}$$

To do this, type

```
z = \{[]; -0.5\}

p = \{0.3; [0.1+i 0.1-i]\}

k = [1; 2]

H = zpk(z,p,k,-1) % unspecified sample time
```

Example 2

Convert the transfer function

Example 3

Create a discrete-time ZPK model from a rational expression in the variable z, by typing

```
z = zpk('z',0.1);

H = (z+.1)*(z+.2)/(z^2+.6*z+.09)

Zero/pole/gain:

(z+0.1) (z+0.2)
```

 $(z+0.3)^2$

Sampling time: 0.1

Algorithm

zpk uses the MATLAB function roots to convert transfer functions and the functions zero and pole to convert state-space models.

See Also

frd Convert to frequency response data models
get Get properties of LTI models
set Set properties of LTI models
ss Convert to state-space models

tf Convert to TF transfer function models

zpkdata Retrieve zero-pole-gain data

zpkdata

Purpose

Quick access to zero-pole-gain data

Syntax

```
[z,p,k] = zpkdata(sys)
[z,p,k] = zpkdata(sys,'v')
[z,p,k,Ts,Td] = zpkdata(sys)
```

Description

[z,p,k] = zpkdata(sys) returns the zeros z, poles p, and gain(s) k of the zero-pole-gain model sys. The outputs z and p are cell arrays with the following characteristics:

- z and p have as many rows as outputs and as many columns as inputs.
- The (i,j) entries z{i,j} and p{i,j} are the (column) vectors of zeros and poles of the transfer function from input j to output i.

The output k is a matrix with as many rows as outputs and as many columns as inputs such that k(i,j) is the gain of the transfer function from input j to output i. If sys is a transfer function or state-space model, it is first converted to zero-pole-gain form using zpk. See Table 11-15, "LTI Properties," on page 11-194 for more information on the format of state-space model data.

For SISO zero-pole-gain models, the syntax

```
[z,p,k] = zpkdata(sys,'v')
```

forces zpkdata to return the zeros and poles directly as column vectors rather than as cell arrays (see example below).

[z,p,k,Ts,Td] = zpkdata(sys) also returns the sample time Ts and the input delay data Td. For continuous-time models, Td is a row vector with one entry per input channel (Td(j)) indicates by how many seconds the jth input is delayed). For discrete-time models, Td is the empty matrix [] (see d2d for delays in discrete systems).

You can access the remaining LTI properties of sys with get or by direct referencing, for example,

```
sys.Ts
sys.inputname
```

Example

Given a zero-pole-gain model with two outputs and one input

```
H = zpk(\{[0];[-0.5]\},\{[0.3];[0.1+i 0.1-i]\},[1;2],-1)
```

Sampling time: unspecified

you can extract the zero/pole/gain data embedded in $\ensuremath{\mathsf{H}}$ with

To access the zeros and poles of the second output channel of H, get the content of the second cell in z and p by typing

```
z{2,1}

ans =
    -0.5000

p{2,1}

ans =
    0.1000+ 1.0000i
    0.1000- 1.0000i
```

See Also

get Get properties of LTI models
ssdata Quick access to state-space data
tfdata Quick access to transfer function data

zpkdata

zpk

Specify zero-pole-gain models

Block Reference

Introduction

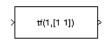
The Control System Toolbox provides the LTI System block for use with Simulink. Its reference page contains the following information:

- The block name and icon
- The purpose of the block
- A description of the block
- The block parameters and dialog box including a brief description of each parameter

Purpose

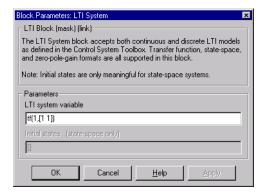
Import LTI System

Description



The LTI System block imports linear, time-invariant (LTI) systems into Simulink.

Dialog Box



LTI system variable

Enter your LTI model. This block supports state-space, zero/pole/gain, and transfer function formats. Your model can be discrete- or continuous-time.

Initial states (state-space only)

If your model is in state-space format, you can specify the initial states in vector format. The default is zero for all states.

LTI System

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