



ÉCOLE POLYTECHNIQUE FÉDÉRALE DE LAUSANNE

CE-1 : Nonparametric Methods

Practical Exercises for System Identification

ME-421 : SYSTEM IDENTIFICATION

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1 CE-1 : Non-parametric Methods

In this first computer exercise we will implement different algorithms learned during the lectures about system identifications and more specifically about non-parametric methods.

The given plant transfer function for different exercises is the following:

$$G(s) = \frac{2 - s}{s^2 + 1.85s + 4} \quad (1)$$

The simulink model adds saturation and noise to this linear model.

1.1 Step response

First, we composed the simulink scheme shown in the Figure 1. It is composed of the transfer function $G(s)$, of a noise with a variance of 0.01 and a sampling time $T_e = 0.1$ [s], and also of a saturation of the input with an upper limit of 0.5 and a lower limit of -0.5. This scheme will be used for all the exercises of this document.

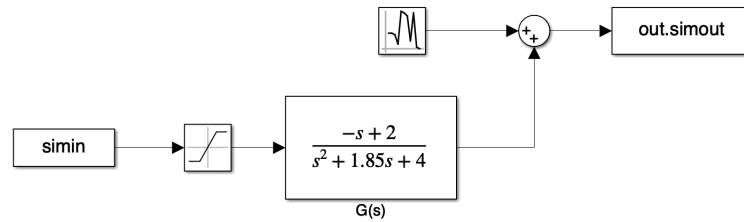


Figure 1
Simulink model

The sample time, $T_e = 0.1$ [s] was imposed on us. We could confirm it was a reasonable choice by using the Nyquist theorem. The Shannon theorem says that the sampling frequency should be at least twice as fast as the nyquist frequency of the system. We use the heuristic that the nyquist frequency is 10 times the bandwidth of the system, and we find a sample time of 0.0848[s] using the Matlab code below in the Listing 1. It is very close to the 0.1[s] given and as taking 10 times the bandwidth is a heuristic, we just want a response of the same order of magnitude. It means $T_e = 0.1$ [s] is a good choice.

```
1 %% Is Te enough ?
2 G = tf([-1 2],[1 1.85 4]);
3
4 fb = bandwidth(G) ; % rad/s
5 fb = fb/(2*pi); % Hz
6
7 f_N = 10*fb; % Nyquist frequency is 10 times the bandwidth
8 fe = 2*f_N; % Nyquist theorem
9
10 Te_th = 1/fe; % sampling time theoretic
11 disp(Te_th);
12
13 % we can see that the natural frequency of the system is way smaller than
14 % than half of the sampling frequency, so Nyquists/Shannon theorem is
15 % verified. Thus the sampling time is reasonable.
```

Listing 1
Matlab code about sample time T_e .

The plots of Figure 2 and Figure 3 display the step and Kronecker delta impulse response of the saturated and noisy system that we computed on Simulink as well as system without saturation or noise.

It is very hard to distinguish the Kronecker impulse response from the noise. Because of the saturation of the system, we cannot just apply a dirac function, we have to use a Kronecker delta function scaled with the saturation limit of the system.

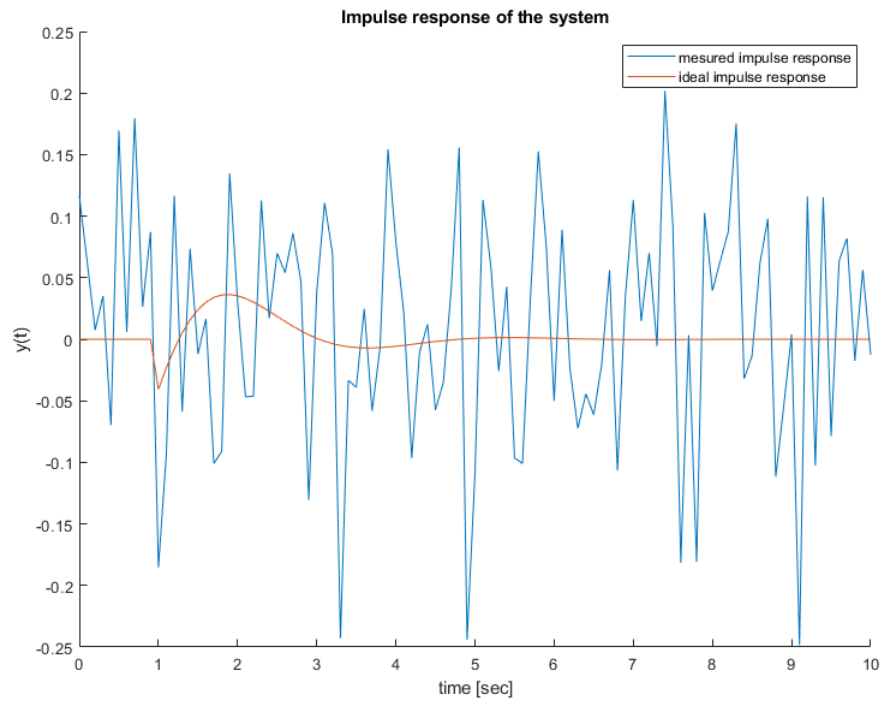


Figure 2
Impulse Response

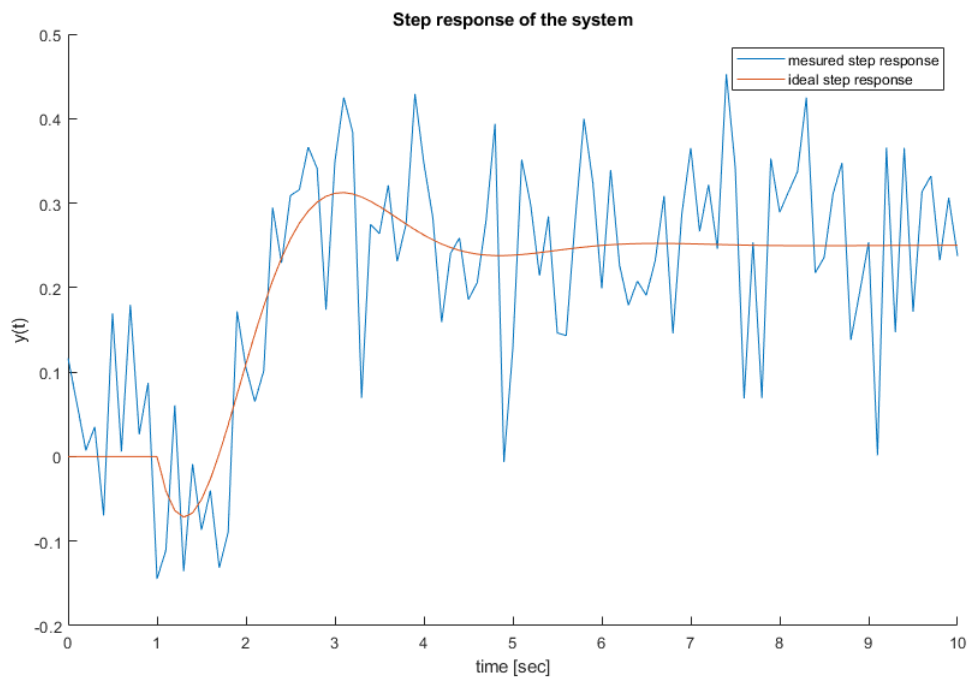


Figure 3
Step Response

In order to obtain these 2 graphs, the Matlab codes in Listing 2 and Listing 3 below have been computed. In the Listing 2, the initialisation of some parameters has been done. In Listing 3, the input vectors for the step and the impulse response are built as well as the graphs.

```

1  sat_up = 0.5; % Upper saturation value
2  Te = 0.1; % Sampling frequency

```

```

3
4 % Time vector
5 tt = (0:Te:10)'; % is a column vector as the exercise demands

```

Listing 2
Matlab code of parameters initialisation

```

1
2 % create values vector
3 values = zeros(size(tt));
4 t_start = 1; % sec (when the step should start)
5 n_start = t_start/Te; % calculate in which sample of the discrete signal to start
   the step
6 values(n_start+1:end) = ones(1,size(tt,1)-n_start)*sat_up;
7
8 % pass to simulink stuct
9 simin.time = tt;
10 simin.signals.values = values;
11
12 % call the simulation
13 out_step = sim('exo1.slx',tt(end));
14
15
16 % ideal response
17
18 G = tf([-1 2],[1 1.85 4]);
19
20 y_step = lsim(G,values,tt);
21
22
23 % plot data
24 hold on
25
26 plot(out_step.simout.Time,out_step.simout.Data);
27 plot(tt,y_step);
28
29 title("Step response of the system");
30 legend("mesured step response","ideal step response")
31 xlabel("time [sec]")
32 ylabel("y(t)")
33
34
35
36 %% do the same for the impulse response
37
38 %create input vector
39 values_dirac = zeros(size(tt));
40 values_dirac(n_start) = sat_up;
41
42 simin.signals.values = values_dirac;
43
44 y_imp = lsim(G,values_dirac,tt);
45 % [y_imp2,tt2] = impulse(G*Te*sat_up);
46
47 % simulate in simulink
48 out_impulse = sim('exo1.slx',tt(end));
49
50 % plot data
51 figure
52 hold on
53 plot(out_impulse.simout.Time,out_impulse.simout.Data);
54 plot(tt,y_imp);
55 % plot(tt2,y_imp2);
56
57
58
59
60 title("Impulse response of the system");
61 legend("mesured impulse response","ideal impulse response")
62 xlabel("time [sec]")
63 ylabel("y(t)")

```

Listing 3
Matlab code to generate step and impulse inputs and responses.

1.2 Auto Correlation of a PRBS signal

In this section, we computed the auto correlation function for a periodic signal in order to use it in the section 1.4 to find the impulse response by correlation approach. You can find below in the Listing 4 the Matlab code of our function to compute the correlation between two periodic signals u and y . Note that it only works if the two signals have the same length.

```

1  function [R,h] = intcor(u,y)
2      % Find M
3      M = size(u,1);
4      if (size(u,1) ~= size(y,1))
5          error("u and y do not have the same size");
6      end
7
8      % Calculate h vector
9      h = 0:M-1;
10
11     % Calculate R(h)
12     R = zeros(size(h));
13     for j = h
14         y_shifted = [y(M-j+1:M);y(1:M-j)]; % We shift y by h: y(k-h)
15         components = u.*y_shifted; % Components to be summed over to get R(h): u
16         % (k)*y(k-h)
17         R(j+1) = sum(components)/M;
18     end
19 end

```

Listing 4

Matlab code about the intcor function

To check this function, we use it to calculate the auto correlation for the signal given by the function $\text{PRBS}(n=6, p=4)$. The code that has been used is the one in the Listing 5. When using the function, we get the following graph in Figure 4:

```

1  u=prbs(6,4);
2  y=u;
3
4  [R,h] = intcor(u,y);
5
6  plot(h,R,'*')
7  title("Autocorrelation function for PRBS (n=6, p=4) ")
8  xlabel("h")
9  ylabel("R_{uu}(h)")

```

Listing 5

Matlab code to test intcor function

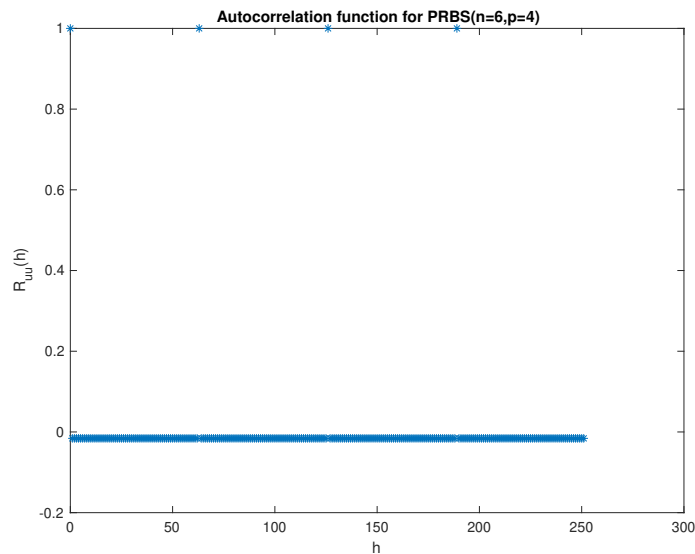


Figure 4

Auto-correlation of a PRBS signal with $n = 6$ (shift register length) and $p = 4$ (number of periods in signal)

We can see that we get the same result as we were to get using the formula 2 given in the system identification course. This formula corresponds to the auto-correlation function of a PRBS signal.

$$R_{uu}(h) = \begin{cases} a^2 & h = 0, \pm M, \pm 2M \dots \\ \frac{-a^2}{M} & \text{elsewhere} \end{cases} \quad (2)$$

Here $M = 2^n - 1$

Indeed, we can see in Figure 4 that we have the value $a^2 = 1$ at each one of the 4 periods and a very small negative value for the rest of the data which is coherent with the formula.

1.3 Impulse response by deconvolution method

In this section we want to compute the Impulse response of the system in simulink using the numerical deconvolution method.

The matrix U is a asymmetric toeplitz matrix created using the vector of inputs. This matrix can be singular, so it is better to suppose that the length of the impulse response is equal to K , with $K \leq N$, so we can solve the problem in the least squares sense.

We have found that $K = 70$ worked well for this system.

For this purpose, we used the formula 3 where Θ_K and U_K are the truncated version of Θ and U .

$$\Theta_K = (U_K^T U_K)^{-1} U_K^T Y \quad (3)$$

```

1 Te = 0.1; %[s]
2 N = 1001; %[-] number of points
3
4
5 tt = (0:Te:(N-1)*Te)';
6
7 sat_up = 0.5;
8 u = rand(size(tt))*sat_up;
9 %no noise system as comparison
10
11 G = tf([-1 2],[1 1.85 4]);
12 y_nonoise = lsim(G,u,tt);
13
14 %simulate system with simulink
15 simin.time = tt;
16 simin.signals.values = u;
17 out_sim = sim('exo3.slx',tt(end));
18 y = out_sim.simout.Data;
19
20 %% deconvolution method (finite impulse response)
21 K = 70;
22
23 % create asymmetric toeplitz matrix
24 r = zeros(1,K);
25 r(1) = u(1);
26 T = toeplitz(u,r);
27
28 % solve least squares
29 g_fir = inv((T')*(T))*((T')*y);
30
31 %% ideal system (no noise, no saturation)
32 sys_d = c2d(G,Te);
33 [g_d,t_d] = impulse(sys_d*Te);

```

Listing 6

Matlab code about the Finite impulse response

This problem can also be solved using the regularisation approach, this method adds a weight λ to the parameters to ensure that we do not try to invert a singular matrix in the least squares algorithm. We have found that $\lambda = 2$ worked well for this system. Increasing λ makes so the optimisation will try to minimise more Θ

The solution to this problem is given by the formula 4 where Φ is the full asymmetric Toeplitz matrix.

We can see the matlab code of linear regression applied to our system in the Listing 7 below.

$$\Theta = (\Phi^T \Phi + \lambda I)^{-1} \Phi^T Y \quad (4)$$

```

1 lambda = 2;
2
3 r = zeros(size(u));
4 r(1) = u(1);
5 T_full = toeplitz(u,r); % full N*N asymmetric toeplitz matrix
6
7 g_reg = inv(T_full'*T_full+lambda*eye(size(T_full)))*(T_full')*y; % solve the least
8 squares problem
9
10 %% plots
11 figure
12 hold on

```



```

12
13 plot(tt(1:K),g_fir)
14 plot(t_d,g_d)
15 plot(tt(1:K),g_reg(1:K))
16
17 err_fir = g_d-g_fir;
18 err_reg = g_d-g_reg(1:K);
19
20 norm_err_fir = norm(err_fir,2)
21 norm_err_reg = norm(err_reg,2)
22
23
24 title("Impulse response")
25 legend("Identified response (FIR)","Ideal response (no saturation, no noise)","
        Identified response (Regularisation)")
26 xlabel("time [s]")
27 ylabel("impulse response [arbitrary units]")

```

Listing 7

Matlab code about the regularisation approach and plots

We can see the comparison of both method: Finite Impulse Response and Regularisation compared to the ideal response, on the Figure 5 below. The 2-norm error of the FIR is 0.1746 and the 2-norm error of the regularisation response is 0.1934. We can conclude that both methods have similar result, however, for this system, FIR is slightly better in term of error minimization and tuning K is more intuitive than tuning λ . To compare to the ideal impulse response, we had to multiply matlab's impulse(G) by the sampling time T_e since the function impulse returns the response to a Kronecker delta of amplitude $\frac{1}{T_e}$.

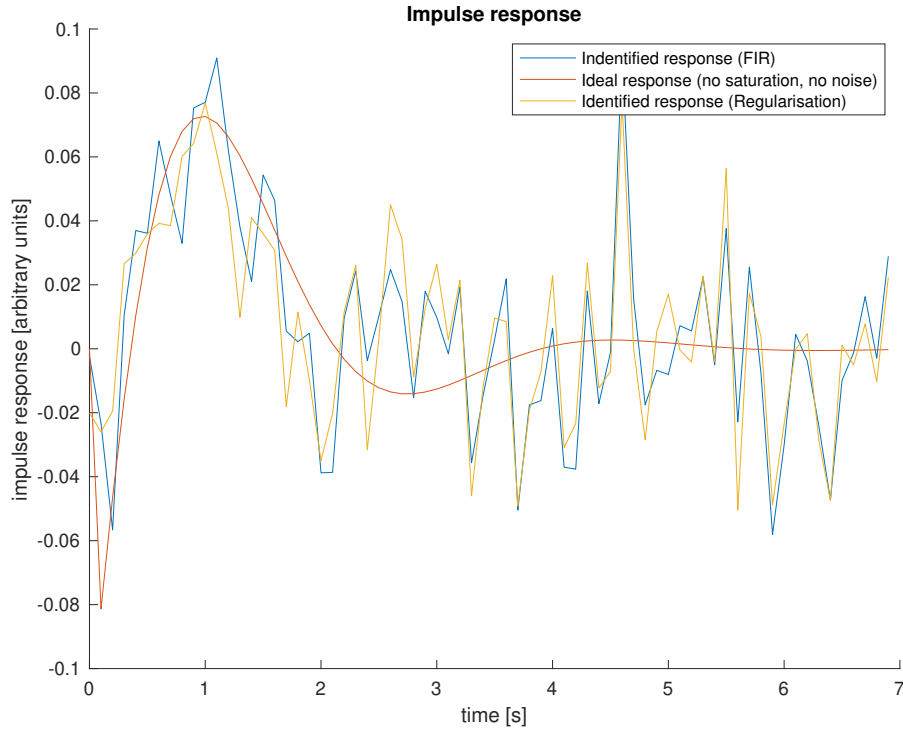


Figure 5

Identified impulse response compared with the true one using two different algorithms (finite impulse response in blue and regularisation in yellow)

1.4 Impulse response by correlation approach

In this part, we want to compute again the Impulse response but this time by using the correlation approach.

Now we do not use the inputs, but the correlation and autocorrelation function to create the toeplitz matrix. The formula is:

$$R_{yu} = U\Theta \quad (5)$$

Where U is now a symmetric toeplitz matrix created with the vector of the auto-correlation of the input R_{uu} , and R_{yu} is the vector with the cross-correlation of the output and input:

$$R_{yu} = \begin{bmatrix} R_{yu}(0) \\ \vdots \\ R_{yu}(M-1) \end{bmatrix} \quad (6)$$

$$R_{uu} = \begin{bmatrix} R_{uu}(0) \\ \vdots \\ R_{uu}(M-1) \end{bmatrix} \quad (7)$$

The measurement noise is way smaller for the correlation approach since in open-loop u and y (the input and output of the plant $G(s)$) are uncorrelated.

Since the signals are periodic, the correlation function can be exactly calculated. This is not exactly the case because of noise, but it is already way better than the normal convolution approach.

We only used the last period of the PRBS signal to ensure that the transient phase was already gone. We have found that including more than a period of the PRBS signal made the toeplitz matrix singular. The results could have been improved by averaging the correlation functions over several periods, but the results obtained seemed sufficient without the need of averaging.

The 2-norm of the error using our correlation function was 0.1996. Using matlab's function, it was 0.1452. So we can conclude that both our function, and matlab's implementation give good results.

```

1  %%
2  sat_up = 0.5; % must fix the maximum amplitude as to not saturate the input
3  Uprbs = prbs(8,8)*sat_up;
4  Te = 0.1;
5  tt = (0:Te:(size(Uprbs,1)-1)*Te)';
6
7  % pass to simulink struct
8  simin.time = tt;
9  simin.signals.values = Uprbs;
10
11 % call the simulation
12 out_step = sim('exo4.slx',tt(end));
13 tt_sim = out_step.simout.Time;
14 y_sim = out_step.simout.Data;
15
16 % pick only the last period of the signal
17 M = 2^(8)-1;
18 Uprbs = Uprbs(7*M + 1:8*M);
19 y_sim = y_sim(7*M + 1:8*M);
20 tt_sim = tt_sim(1:M);
21
22 R_uu_intcor = intcor(Uprbs,Uprbs);
23 R_yu_intcor = intcor(y_sim,Uprbs);
24
25 U_toeplitz_intcor = toeplitz(R_uu_intcor);
26
27 g_k_intcor = pinv(U_toeplitz_intcor)*(R_yu_intcor');
28
29 %% now with matlab
30
31 R_uu_matlab = xcorr(Uprbs,Uprbs);
32 R_yu_matlab = xcorr(y_sim,Uprbs);
33
34 %keep only the positive part of the correlations functions
35

```

```

36 R_yu_matlab = R_yu_matlab((end+1)/2:end);
37 R_uu_matlab = R_uu_matlab((end+1)/2:end);
38
39 U_toeplitz_matlab = toeplitz(R_uu_matlab);
40 g_k_matlab = pinv(U_toeplitz_matlab)*R_yu_matlab;
41
42 plot(R_uu_matlab)
43
44 figure
45 % figure
46
47
48 %% exact reponse of the system
49 G = tf([-1 2],[1 1.85 4]);
50 G = c2d(G,Te);
51 g_theory = impulse(G,tt_sim)*Te;
52
53 %% plots
54 hold on
55 plot(tt_sim,g_k_intcor)
56 plot(tt_sim,g_k_matlab)
57 plot(tt_sim,g_theory)
58
59 err_intcor = g_theory-g_k_intcor;
60 err_matlab = g_theory-g_k_matlab;
61
62 err_norm_intcor = norm(err_intcor,2)
63 err_norm_matlab = norm(err_matlab,2)
64
65 title("Impulse response g(t)")
66 xlabel("discrete time k")
67 legend("Impulse response with intcor()", "Impulse response with xcorr()", "Theoretical impulse (no noise, no saturation)")

```

Listing 8

Matlab code about computing the impulse response using intcor and xcorr

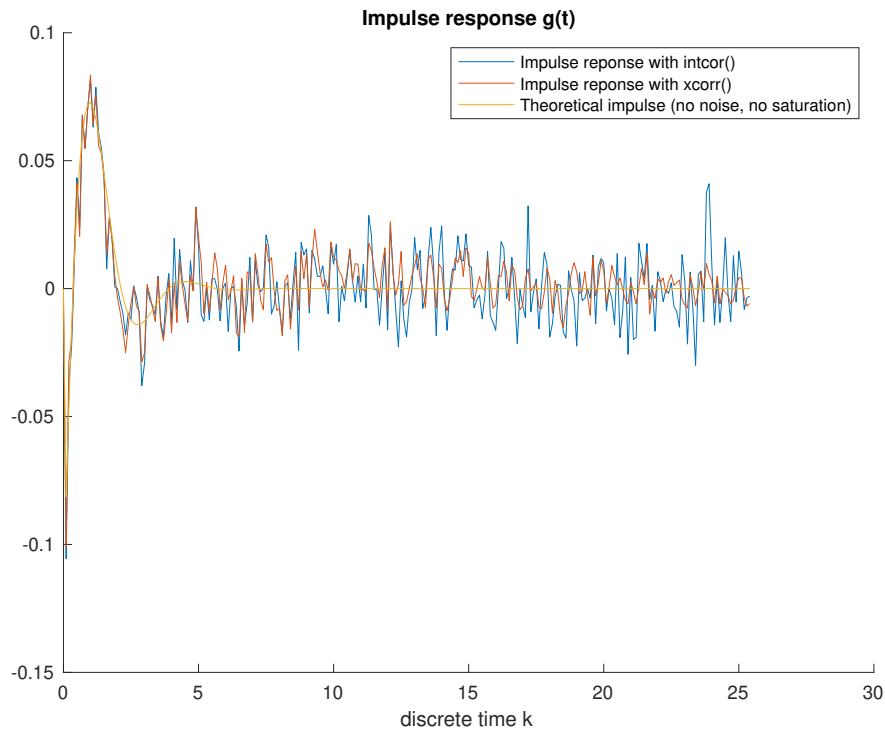


Figure 6

Identified impulse responses by correlation approach compared with the true one.

1.5 Frequency domain Identification (Periodic signal)

For the PRBS, we chose $p = 16$ and to get more than 2000 points, we chose $n = 7$, that gives us 2032 points in the PRBS signal. We average the fourier coefficients over the multiple periods, but skipping the first period to ignore the transient response of the system.

We can see that the bode diagram of the identified system is quite similar to the ideal bode diagram. This is because we averaged the fourier coefficients over several periods and this reduces the measurement errors. We have no truncation error since the signals are periodic, and we respect the Shannon's theorem so we should not get any considerable sampling error.

```
1  clc,close all,clear all
2  %% 1
3
4  sat_up = 0.5; % must fix the maximum amplitude as to not saturate the input
5  p = 16;
6
7  N_wanted = 2000;
8  n_min = ceil(log2((N_wanted+1)/p)); %minimum shift register length
9
10 Uprbs = prbs(n_min,p)*sat_up;
11 size(Uprbs)
12 Te = 0.1;
13 tt = (0:Te:(size(Uprbs,1)-1)*Te)';
14
15 % pass to simulink struct
16 simin.time = tt;
17 simin.signals.values = Uprbs;
18
19 % call the simulation
20 out_step = sim('exo5.slx',tt(end));
21 tt_sim = out_step.simout.Time;
22 y_sim = out_step.simout.Data;
23
24 %% 2
25
26 N = size(Uprbs,1);
27
28 Uprbs_fft_avg = 0;%initialiation
29 y_fft_avg = 0;%initialiation
30 N_one_period = N/p;
31
32 number_periods_to_ignore = 1;
33
34 for i = (1+number_periods_to_ignore*N_one_period):N_one_period:N
35     Uprbs_period = Uprbs(i:i+N_one_period-1);
36     Uprbs_fft_avg = (fft(Uprbs_period) + Uprbs_fft_avg);
37
38     y_period = y_sim(i:i+N_one_period-1);
39     y_fft_avg = (fft(y_period) + y_fft_avg);
40 end
41 Uprbs_fft_avg = Uprbs_fft_avg/(p-number_periods_to_ignore);
42 y_fft_avg = y_fft_avg/(p-number_periods_to_ignore);
43
44
45 %% 3
46
47 f_s = 1/Te;
48 omega_s = 2*pi*f_s;
49
50 omega_vec = (omega_s./N_one_period).*(0:(N_one_period-1));
51
52 G = y_fft_avg./Uprbs_fft_avg;
53
54 %% 4
55 sys_f = frd(G(1:floor(end/2)),omega_vec(1:floor(end/2)),Te);
56
57 %%
58 G = tf([-1 2],[1 1.85 4]);
59 G = c2d(G,Te);
60
61
62 bode(sys_f,G)
63 legend("identified system","ideal system")
```

Listing 9

Matlab code about Fourier analysis

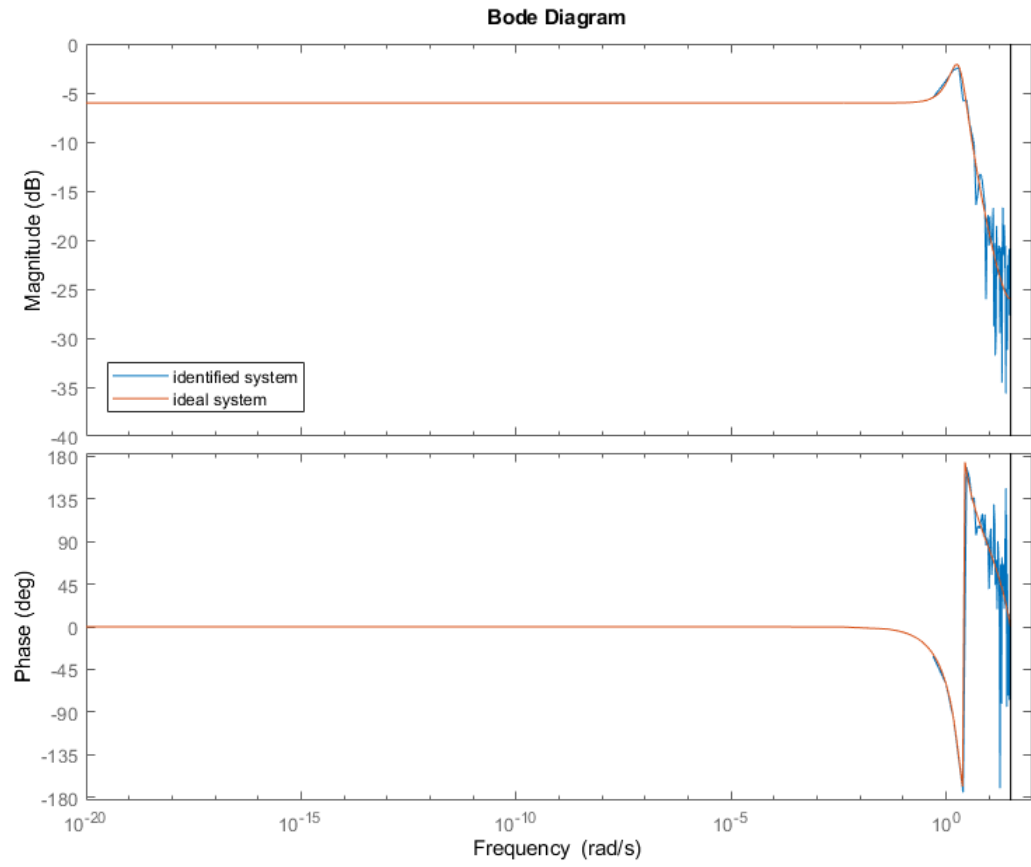


Figure 7
Bode diagram of the identified model [blue] compared with the true one [orange].

1.6 Frequency domain Identification (Random signal)

We have chosen to apply binary random signal to the system to deliver the maximum amount of energy on the signal and thus increase the signal to noise ratio.

We have chosen uniform white noise. The important part is that its white noise, so that the we have a flat spectrum to excite as much frequencies as possible.

For the first identification, we use a biased correlation estimate because the unbiased estimate has a high variance for large $|k|$. Using a biased estimate is equivalent as using an unbiased one with a triangular window of size N .

To reduce the truncation error, we do a second identification using a Hann window. It is important to note that we only use half of the window for the positive part or the correlation functions.

Because we are using the window, we can use the unbiased estimate of the correlation functions.

We have found that a window size of $M = 300$ gave us good results (the actual entire window has size $2M$). Increasing M increases the resolution but also increases the spectrum leakage problem. For the Hann window, the secondary lobes are very small so this is not a problem.

Using a $M < N$ is equivalent of padding the rest of the window with zeros. And in our case, $M < N$ gave a better result than for $M = N$. This is again because of the high variance for large $|k|$ of the unbiased correlation estimate.

For the last identification, we have split the data in $m = 10$ groups and averaged the power spectrum functions ϕ_{uu} and ϕ_{yu} to reduce the variance of the estimated parameters. We have also used the Hann window on each segment to reduce the truncation error.

We can see from our results that while the first identification has worked, it has a lot of truncation error that is significantly removed by using the Hann window. The second identification using the window is thus way better. The last identification averaging the power spectrum functions seems to help a bit more but most of the noise was coming from truncation so the second and third identification are quite similar.

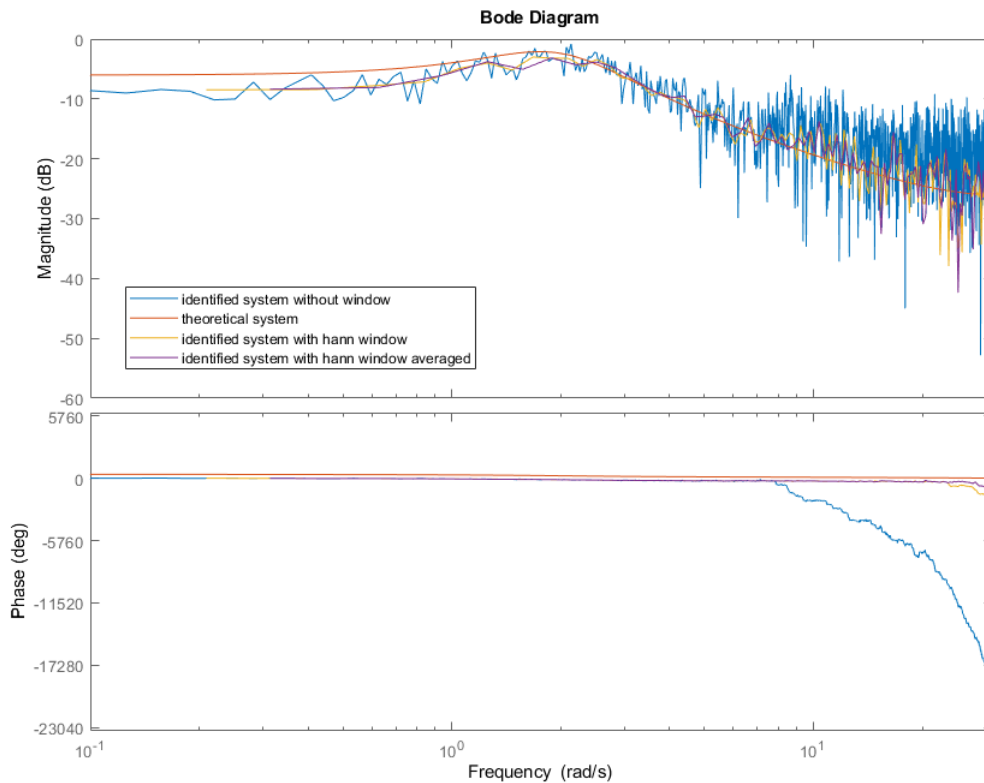


Figure 8
Bode diagram of the identified models and the ideal model.

```
1 clc,close all,clear all
2 %% exo 6
```

```

3 %% 1
4 N = 2000;
5 sat_up = 0.5;
6 Te = 0.1;
7
8 U = rand(N,1)-0.5;
9
10 U(U<0)=-sat_up;
11 U(U>0)=sat_up;
12
13 tt = (0:Te:(size(U,1)-1)*Te)';
14
15 % call simulink
16 simin.time = tt;
17 simin.signals.values = U;
18 out_step = sim('exo6.slx',tt(end));
19 tt_sim = out_step.simout.Time;
20 y_sim = out_step.simout.Data;
21
22
23 %% 2
24
25 Ruu = xcorr(U,U,'biased');
26 Ryu = xcorr(y_sim,U,'biased');
27
28 %keep only the positive part of the correlations functions
29 Ruu = Ruu((end+1)/2:end);
30 Ryu = Ryu((end+1)/2:end);
31
32 phi_uu = fft(Ruu);
33 phi_yu = fft(Ryu);
34
35 f_s = 1/Te;
36 omega_s = 2*pi*f_s;
37 omega_vec = (omega_s./N).*(0:(N-1));
38
39
40 G_id = phi_yu./phi_uu;
41
42 sys_id = frd(G_id(1:floor(end/2)),omega_vec(1:floor(end/2)),Te);
43
44 % real system (no noise, no saturation)
45 G = tf([-1 2],[1 1.85 4]);
46 G = c2d(G,Te);
47 %% 3
48
49 M = 300;
50 window = hann(2*M);
51 window = window((end)/2 + 1:end);
52
53 Ruu = xcorr(U,U,'unbiased');
54 Ryu = xcorr(y_sim,U,'unbiased');
55
56 %keep only the positive part of the correlations functions
57 Ruu = Ruu((end+1)/2:end);
58 Ryu = Ryu((end+1)/2:end);
59
60
61 phi_uu_windowed = fft(Ruu(1:M).*window);
62 phi_yu_windowed = fft(Ryu(1:M).*window);
63 G_windowed = phi_yu_windowed./phi_uu_windowed;
64
65 omega_vec_windowed = (omega_s./M).*(0:(M-1));
66
67 sys_id_windowed = frd(G_windowed(1:floor(end/2)),omega_vec_windowed(1:floor(end/2)),Te);
68
69 %% 4
70
71 m = 10;
72 samples_per_group = N/m;
73
74 window = hann(2*samples_per_group);
75 window = window((end)/2 + 1:end);
76
77 phi_uu_avg = zeros(samples_per_group,1);%initialiation

```

```

78 phi_yu_avg = zeros(samples_per_group,1);%initialiation
79
80 for i = 1:samples_per_group:N
81     U_period = U(i:i+samples_per_group-1);
82     y_period = y_sim(i:i+samples_per_group-1);
83
84
85     Ruu = xcorr(U_period,U_period,'unbiased');
86     Ryu = xcorr(y_period,U_period,'unbiased');
87
88     Ruu = Ruu((end+1)/2:end);
89     Ryu = Ryu((end+1)/2:end);
90
91     phi_uu_avg = fft(Ruu.*window) + phi_uu_avg;
92     phi_yu_avg = fft(Ryu.*window) + phi_yu_avg;
93 end
94
95 phi_uu_avg = phi_uu_avg/m;
96 phi_yu_avg = phi_yu_avg/m;
97
98 G_windowed_avg = phi_yu_avg./phi_uu_avg;
99 omega_vec_windowed_avg = (omega_s./samples_per_group).*(0:(samples_per_group-1));
100
101 sys_id_windowed_avg = frd(G_windowed_avg(1:floor(end/2)),omega_vec_windowed_avg(1:
    floor(end/2)),Te);
102
103
104 %% 5
105
106 figure
107 bode(sys_id,G,sys_id_windowed,sys_id_windowed_avg);
108 legend("identified system without window","theoretical system","identified system
    with hann window","identified system with hann window averaged")
109 xlim([10^-1 inf])

```

Listing 10

Matlab code about frequency domain identification