

# **Chapter VII**

## **Practical Aspects of System Identification**

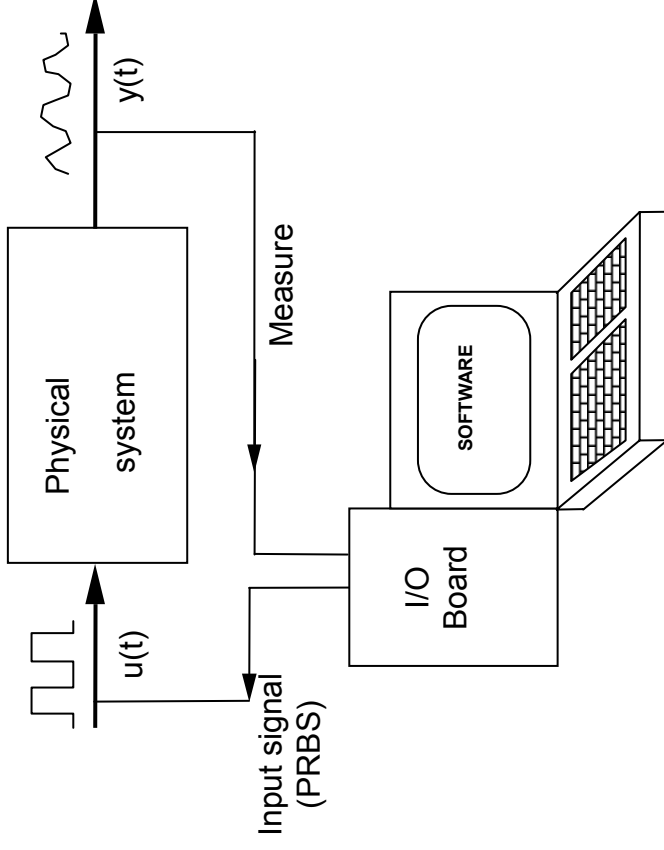
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# **Chapter 7. Practical Aspects of System Identification**

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# Input/Output Data Acquisition

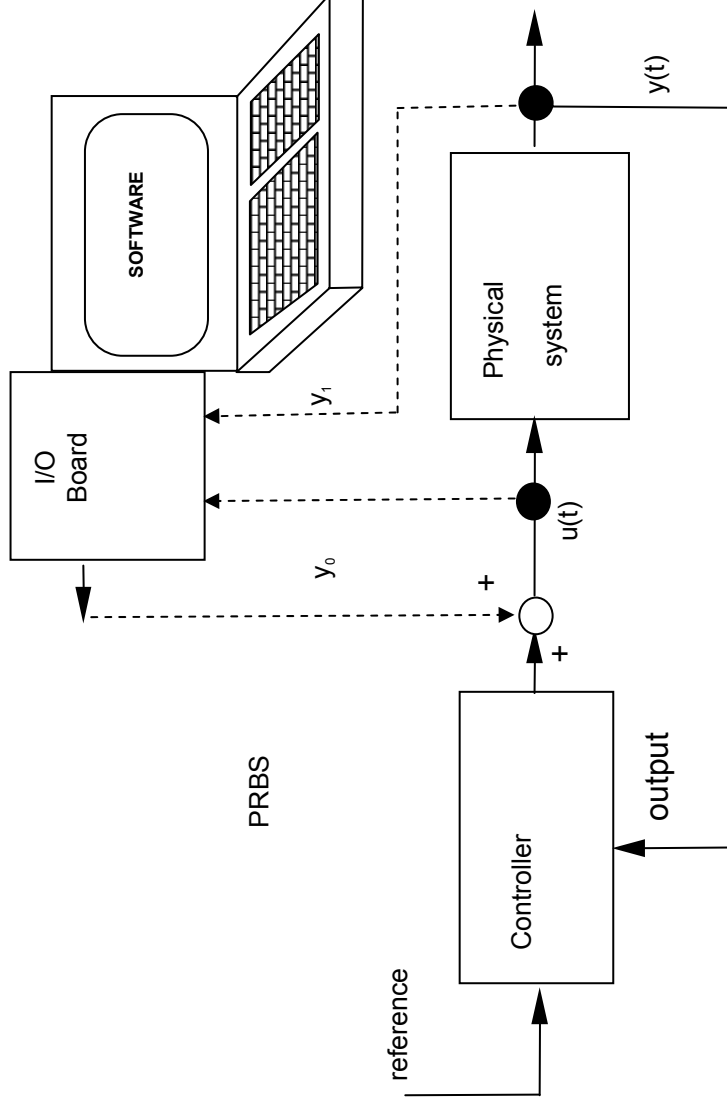
Plant operated in open loop



Plant directly excited by a PRBS signal

# Input/Output Data Acquisition

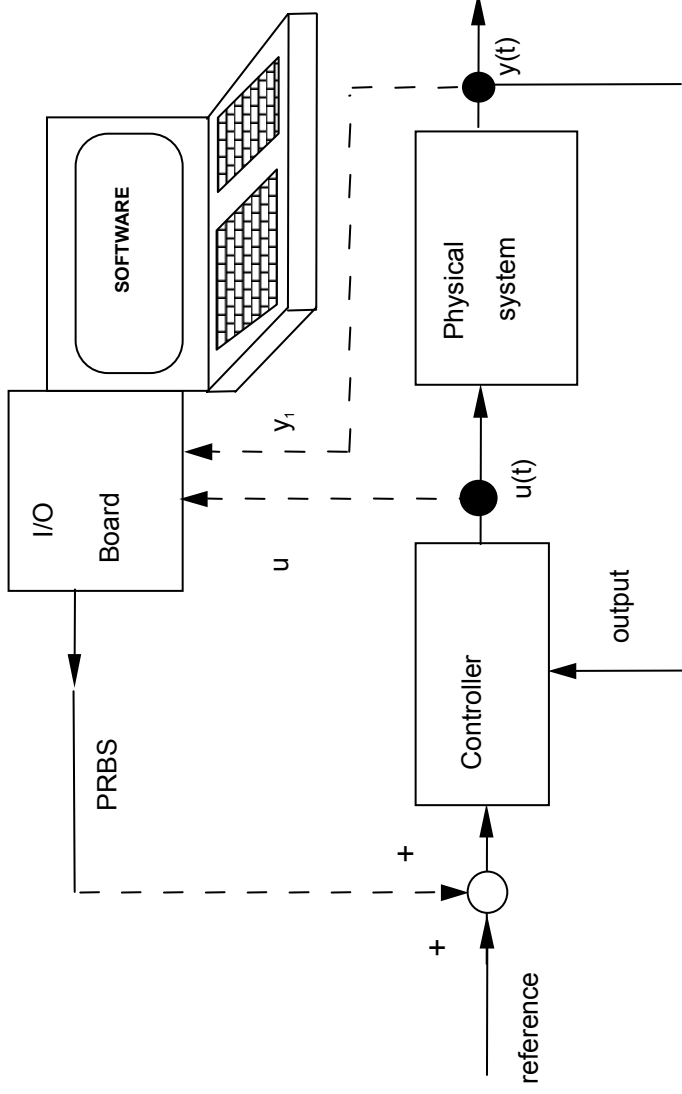
Plant operated in closed loop with excitation added to the controller output



- PRBS superposed to the controller output
  - Identification of the transfer function between  $y_0$  and  $y$
- Use of a controller with integral action but limited proportional and derivative actions*

# Input/Output Data Acquisition

Plant operated in closed loop with excitation added to the reference



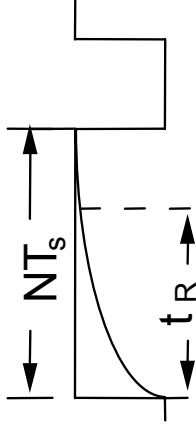
- PRBS superposed to the reference
  - Identification of the transfer function between  $y_0$  and  $y$
- Use of a controller with integral action but limited proportional and derivative actions*

## Sizing of the PRBS signal

Good identification of the static gain:

*Approach 1 (choice of  $N$ )*

$$t_{\text{im}} = T_s \cdot N > t_R$$



But on the other hand :  $(2^N - 1)T_s = L < \text{experiment length}$

$$N \rightarrow NT_s \geq t_R ; (2^N - 1)T_s \leq \text{experiment length}$$

*Approach 2 (choice of  $N$  and of the frequency divider  $f_{PRBS}$ )*

$$f_{PRBS} = \frac{f_s}{p} ; p = 1, 2, 3, \dots$$

$$N, p \rightarrow pNT_s > t_R ; p(2^N - 1)T_s \leq \text{experiment length}$$

**Magnitude : 0.5%(0.15%) to 10%(5%) of the input steady state value**

## Anti-aliasing filtering

- useful frequencies until  $0.5 f_s$
- frequencies beyond  $0.5f_s$  modify the spectrum of the sampled signal (spectrum overlapping)

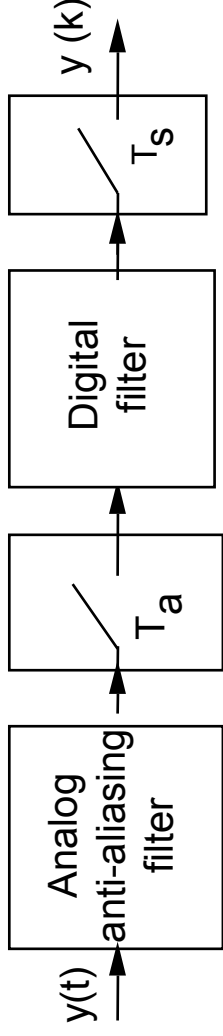
Consequence : bad identification

An anti-aliasing filter is necessary

- Solution 1 : analog filters
- Solution 2 : Over-sampling (for very slow frequency,  $f < 2\text{Hz}$ )

## Oversampling

- In many digital control systems (DCS), the data acquisition frequency is significantly higher than that used in the control loops.
- Very small low sampling frequency ( $< 2\text{Hz}$ )



A/D converter (acquisition frequency)      under-sampling  
( $T_s = n \cdot T_a$ )

$$f_a = n f_s$$

$f_a$  = acquisition frequency

$n$  = integer frequency multiple

$f_s$  = sampling frequency of the control loop



## Over-sampling

Digital anti-aliasing filter

For  $n > 3$  a moving-average filter is sufficient

$$y_f(t) = \frac{y(t) + y(t-1) + \dots + y(t-n+1)}{n}$$

Sizing of the PRBS signal

*The frequency divider used for the PRBS should be a multiple of  $n$ , in order that the final sampling frequency  $f_s$  is still an integer multiple of the clock frequency of the PRBS.*

PRBS clock frequency :

$$f_{PRBS} = \frac{1}{p} f_a$$

Final sampling frequency:

$$f_s = \frac{1}{n} f_a = \frac{p}{n} f_{PRBS}$$

**Warning:**  $(p/n)$  should be an integer

### Elimination of the DC Component

It is necessary to work with signal variations (dynamics models)

If the DC component is not eliminated then: *bias on the identified parameters*

Stationary (or almost stationary) continuous components

- a) Computation of the Mean Value (M.V.) from the I/O data set.
- b) Elimination of the Mean Value (M.V.) from the I/O data set, and a new centered file is created.

Non-stationary continuous components

The I/O signals are replaced by their corresponding variations (eventually filtered).

$$y'(t) = \frac{y(t) - y(t-1)}{1 + f_1 q^{-1}} \quad ; \quad u'(t) = \frac{u(t) - u(t-1)}{1 + f_1 q^{-1}} \quad -0,5 \leq f_1 \leq 0$$

## Signal conditioning

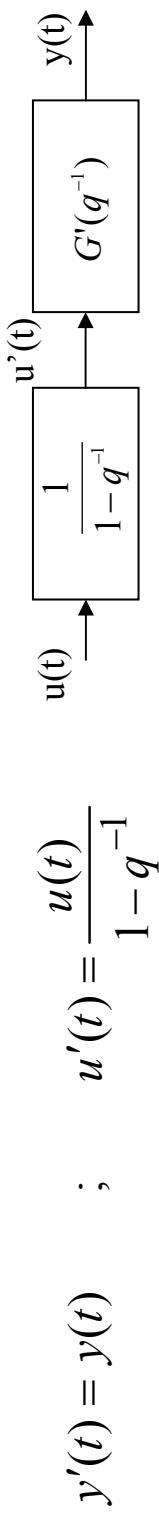
### Identification of a plant containing a pure integrator

If the plant is:  $G(q^{-1}) = \frac{1}{1-q^{-1}}G'(q^{-1})$

It is easier to identify  $G'$

#### Method 1

- The input is replaced with its integral, the output is unchanged



#### Method 2

- The output is replaced with its variations, the input is unchanged



## Signal conditioning

### Scaling of the I/O data set

Observations vector:  $\Phi(t) = [-y(t), -y(t-1), \dots, u(t), u(t-1), \dots]$

Adaptation gain matrix:  $F(t)^{-1} = \sum_{i=1}^t \Phi(i-1)\Phi^T(i-1) + \frac{1}{\delta}I$  ;  $\delta \ll 1$

If the magnitude of  $u(t), u(t-1) \dots$  is very different from that of  $y(t), y(t-1)$ , the gain matrix is not equilibrated, leading to very different convergence speed for the parameter  $\hat{\mathbf{a}}_i(t)$  and  $\hat{\mathbf{b}}_i(t)$

The I/O signals should be scaled (by multiplying, if necessary, both  $u(t)$  and  $y(t)$ )

Once the identification is performed, the coefficients  $\hat{b}_i(t)$  obtained must be re-scaled in order to obtain the good static gain

$$u_F = \alpha u \quad y_F = \beta y \quad y = Gu; \quad y_F = G_F u_F \Rightarrow \beta y = G_F \alpha u$$

$$G = \frac{\alpha}{\beta} G_F$$

## Selection of the model complexity

(if no algorithm for the automatic system order estimation is available)

The model: 
$$A(q^{-1})y(t) = q^{-d}B(q^{-1})u(t) + C(q^{-1})e(t)$$

$$n_A = ? \quad n_B = ? \quad n_C = ? \quad d = ?$$

« *a priori* » choice of  $n_A$

Two cases :

**1)** industrial plant (temperature regulation, flow, concentration, etc.).

In general for these types of systems it holds :

$$n_A \leq 3$$

$n_A = 2$  : very common value, (good initialization )

**2)** Electromechanical systems.

$n_A$  results from the structural analysis of the system.

*Example* : flexible transmission with 2 resonant modes.

One selects  $n_A = 4$ , since a 2<sup>nd</sup> order is needed for modeling each resonant mode.

## Selection of the model complexity

Initial value for  $\mathbf{d}$  and  $\mathbf{n}_B$

- If no knowledge about the delay :  $d=0$  (or  $d=d_{\min}$  if a lower limit is known)
- The first coefficients of  $B$  are very small if  $d=0$  and a delay is present
- One chooses :  $n_B \geq d_{\max} + 2$

Estimation of  $\mathbf{d}$

- Identification by RLS

$$B(q^{-1}) = b_1q^{-1} + b_2q^{-2} + b_3q^{-3} + \dots; \quad \text{If } |b_1| < 0.15|b_2| \rightarrow b_1 \approx 0 \rightarrow d = 1$$

$$B(q^{-1}) = q^{-1}(b_2q^{-1} + b_3q^{-2} + \dots) \rightarrow d = 1$$

$$\text{If } |b_i| < 0.15|b_{i+1}|; \quad i = 1, 2, \dots, d \rightarrow \text{delay } d$$

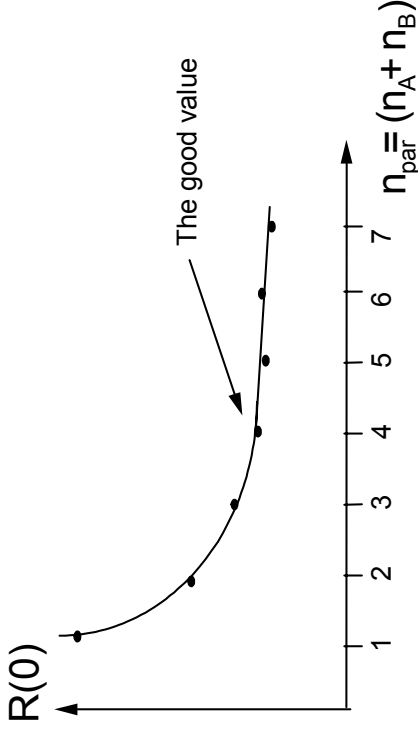
- This test can be completed by a visual validation with a step response
- The better estimation for  $d$  is done with the « validated » model

Choice for  $\mathbf{n}_C$ : in general  $n_C = n_A$

## Determination of $n_A$ max and $n_B$ max

Objective : to find the simplest model which validates

Test: RLS, Evolution of  $R(0) = E(\varepsilon^2)$  as a function of the order



Consider  $nA' = nA + I$ ,  $nB$  and the corresponding variance of the residual errors  $R'(0)$ .

If:  $R'(0) \geq 0.8R(0)$  then it is not advisable to increase the order  
( same for  $n_B' = n_B + I$  )

- Try different identification methods for finding a valid model without changing the orders of  $n_A$  and  $n_B$
- If no valid model is obtained then increase the orders of  $n_A$  and/or  $n_B$  (but not simultaneously)

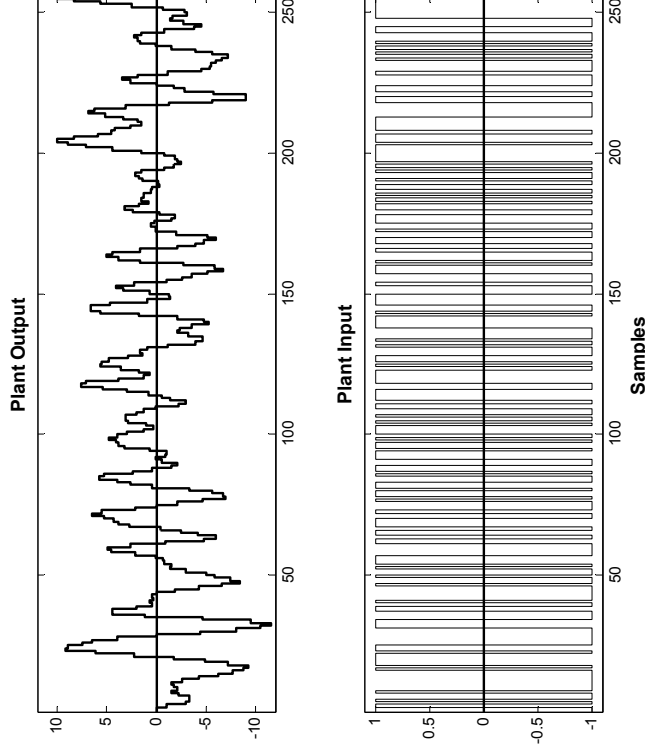
## File for testing the algorithms (simulation)

File T0 (no noise added to the measure):

$$A(q^{-1})y(t) = q^{-1}B(q^{-1})u(t)$$

$$A(q^{-1}) = 1 - 1.5q^{-1} + 0.7q^{-2}$$

$$B(q^{-1}) = 1q^{-1} + 0.5q^{-2}$$



File T0

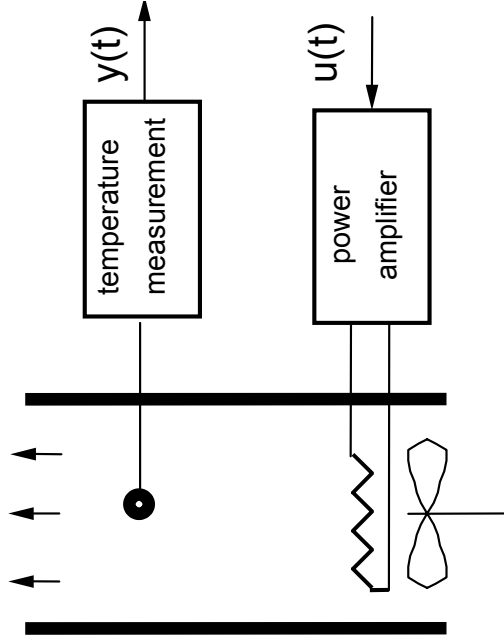
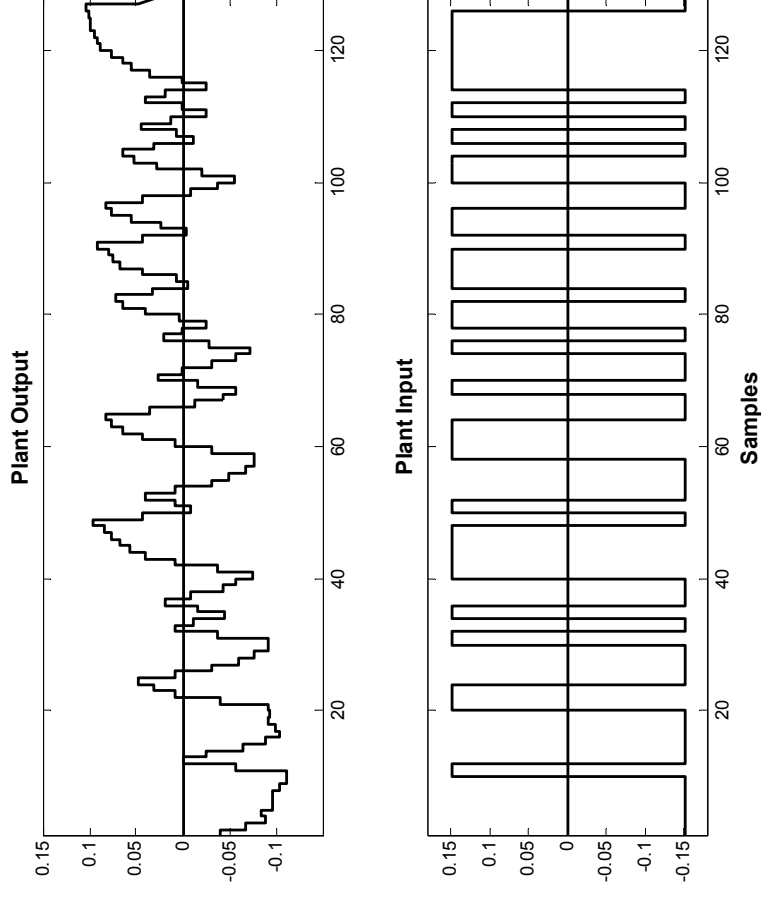
File T1 (with noise added to the measure):

$$A(q^{-1})y(t) = q^{-1}B(q^{-1})u(t) + C(q^{-1})e(t)$$

*Available on the book website*



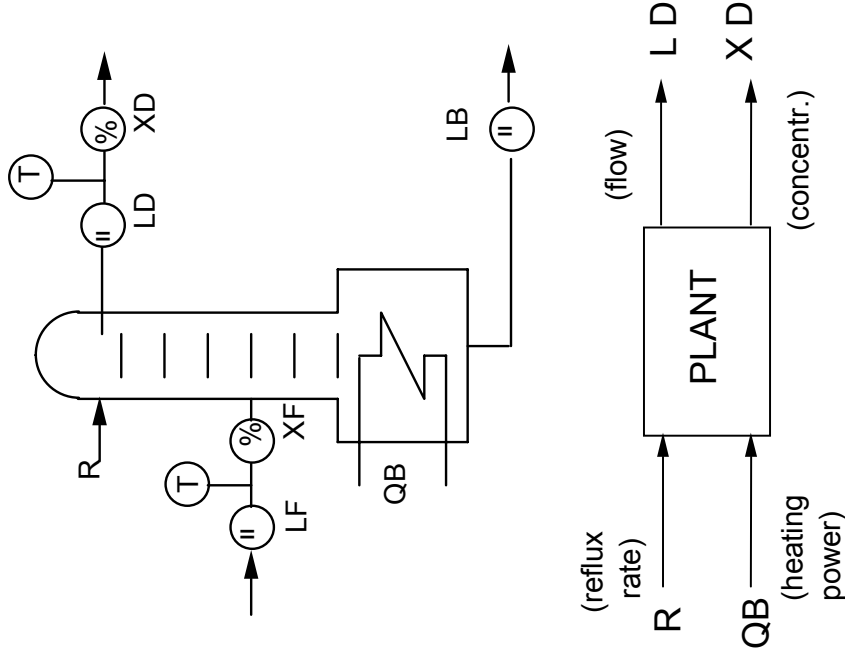
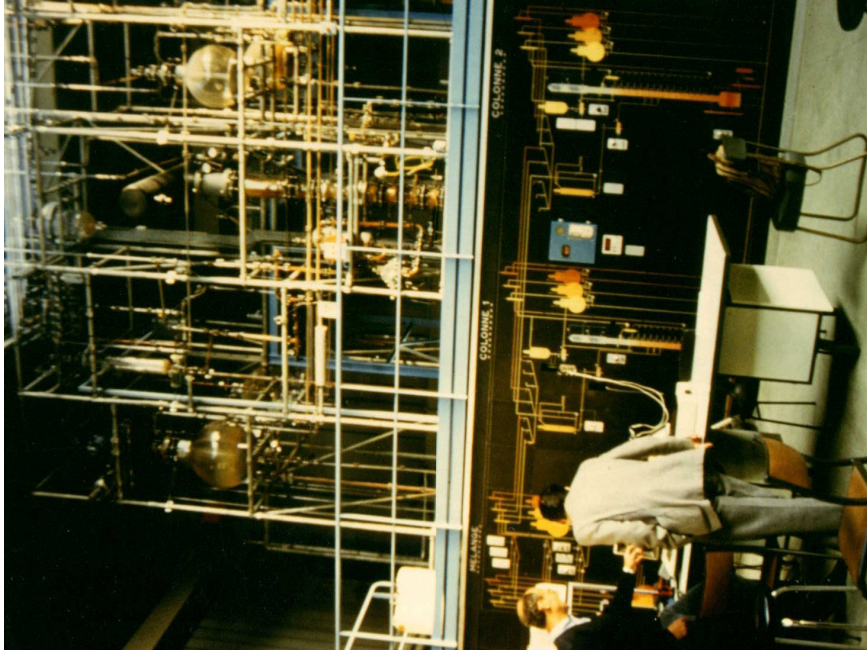
# Air heater



File : Aero.c (see book website)

*See details of the results in the book (pg. 296 - 300)*

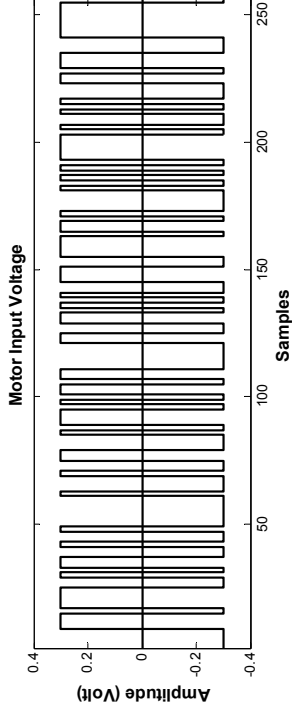
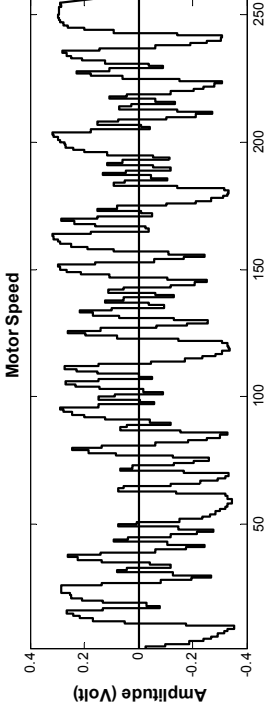
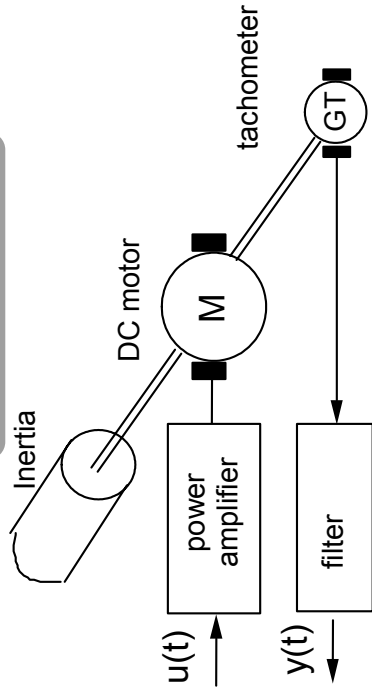
# Distillation column



File : QXD (see book website)    Input : QB    Output: XD

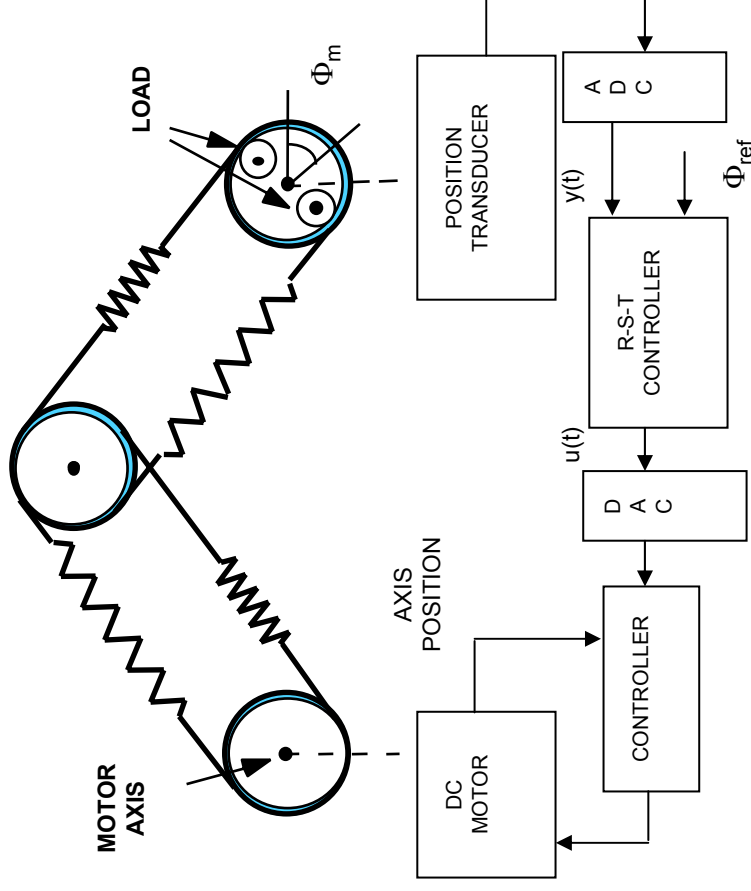
*See details of the results in the book (pg. 300 - 305)*

# DC Motor



File : Mot3.c (see book website)  
See details of the results in the book (pg. 305 - 309)

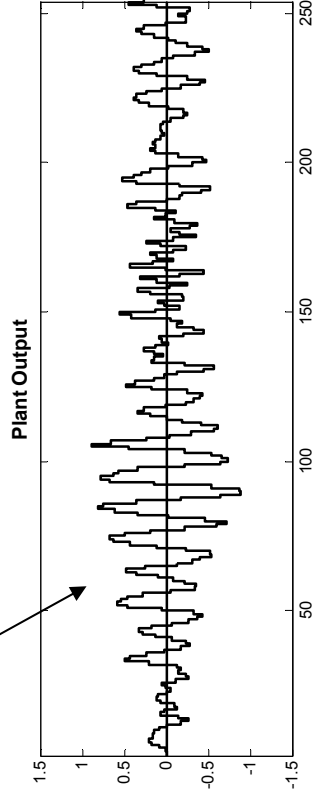
# Flexible transmission



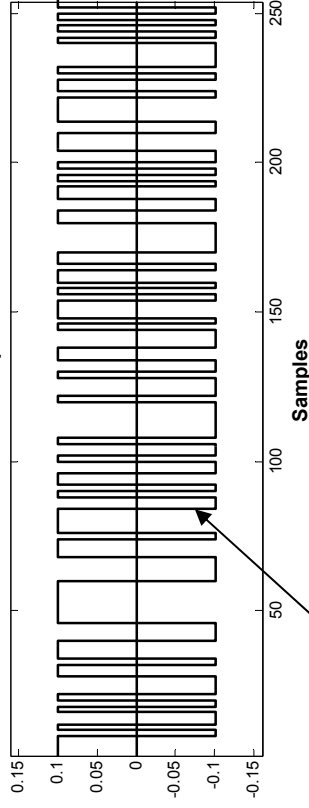
File : Poulbo1.C (see book website)

# Flexible transmission identification

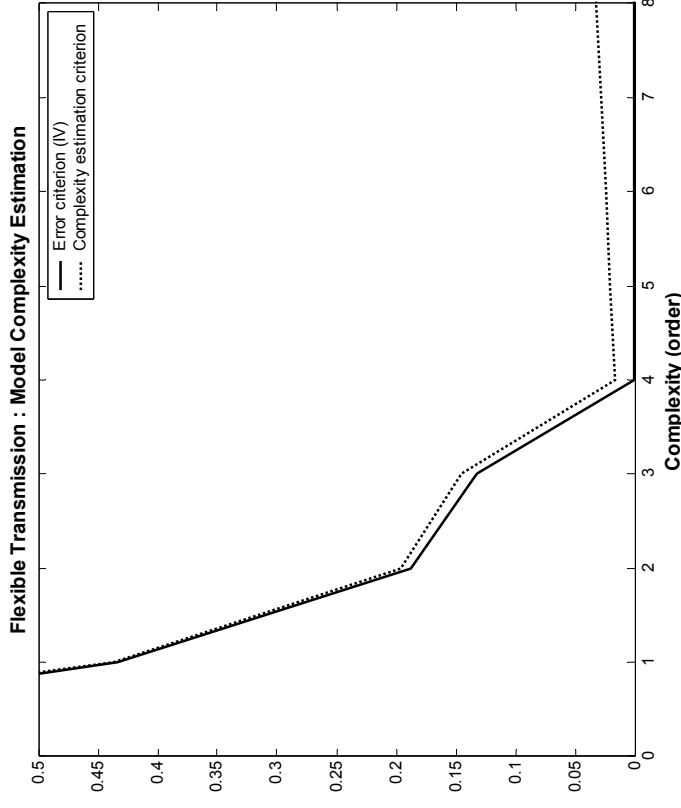
Measured position (y)



Plant Input



Excitation signal (u)  
(PRBS)



Complexity estimation  
Order  $n = \max(n_A, n_B + d)$

*Remark* : The system is characterized by two resonant modes

# Flexible transmission identification

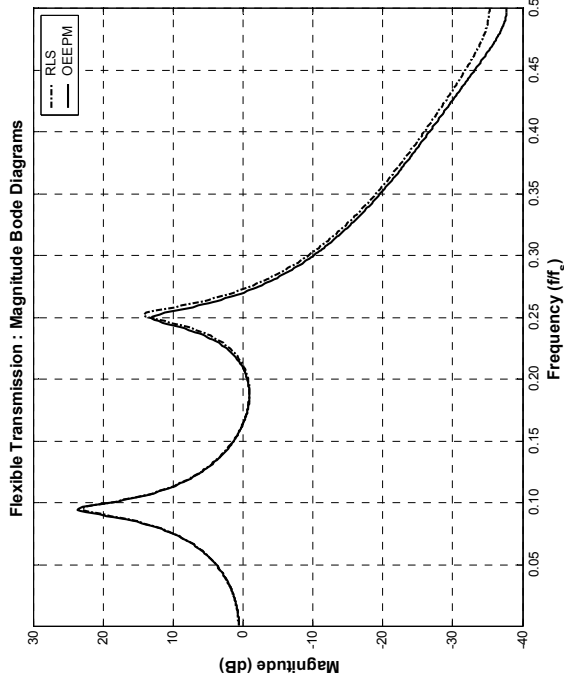
S=1 M=1 (RLS) A=1 FILE:POULBO1.C NS=254 DELAY D=2  
 COEFFICIENTS OF POLYNOMIAL A: A(1) = -1.5748  
 A(2) = 1.8329  
 A(3) = -1.4784  
 A(4) = 0.8895  
 COEFFICIENTS OF POLYNOMIAL B: B(1) = 0.3010  
 B(2) = 0.4181  
 VALIDATION TEST: Whiteness of the residual error  
 System variance: 0.3317 Model variance: 0.1053 Error variance R(0): 0.0007  
 NORMALIZED AUTOCORRELATION FUNCTIONS  
 Validation Criterion: Theor. Val.: |RN(i)| 0.136, Pract. Val.: |RN(i)| 0.15  
 RN(0) = 1.0000 → RN(1) = -0.5727 ←  
 → RN(2) = 0.2360 ← RN(3) = -0.0475  
 RN(4) = -0.0158

R.L.S.

S=3 M=3 (OEPM) A=1 FILE:POULBO1.C NS=254 DELAY D=2  
 COEFFICIENTS OF POLYNOMIAL A: A(1) = -1.60955  
 A(2) = 1.87644  
 A(3) = -1.49879  
 A(4) = 0.88574  
 COEFFICIENTS OF POLYNOMIAL B: B(1) = 0.30530  
 B(2) = 0.39430  
 COEFFICIENTS OF POLYNOMIAL C: C(1) = -0.67530 C(2) = 0.2283  
 C(3) = -0.0653 C(4) = -0.0585  
 VALIDATION TEST: Whiteness of the residual error  
 System variance: 0.1061 Model variance: 0.1055 Error variance R(0):  
 0.0004  
 NORMALIZED AUTOCORRELATION FUNCTIONS  
 Validation Criterion: Theor. Val.: |RN(i)| 0.136, Pract. Val.: |RN(i)| 0.15  
 RN(0) = 1.0000  
 RN(1) = -0.0425  
 RN(2) = 0.0959  
 RN(3) = -0.0563  
 RN(4) = -0.0407

O.E.E.P.M.

Good  
Validation



Frequency responses of  
the models identified

For more details  
see book (pg. 309 – 314)

## A few concluding remarks

- The acquisition should be done by applying a signal rich in frequencies in the bandwidth of the system
- The PRBS is a basic signal that must be sized according to the application considered (as a function of the test time length and the time response)
- Attention to the filtering before the acquisition (anti-aliasing filtering)
- The I/O signals must be centered !
- The models identified must be validated!

### **This can help:**

*Matlab (Scilab) functions for generation of PRBS, complexity estimation, parameter identification and model validation are available on the book website*