





# **Communication Systems and Protocols**

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#### **Communication Systems and Protocols**

Date: 22.07.2013 Name: «Vorname» «Nachname» Matriculation ID: «Matrikelnummer» ID-No.: «LaufNr» Institut für Technik der Informationsverarbeitung (ITIV) Prof. Dr.-Ing. Klaus Müller-Glaser Prof. Dr.-Ing. Dr. h. c. Jürgen Becker Prof. Dr. rer. nat. Wilhelm Stork

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#### Prerequisites for the examination

#### Aids

- Allowed aids for the examination are writing utensils, a ruler, a non-programmable calculator and a single sheet of A4 paper with self- and hand-written notes. Writing may be on a single side of the paper only. The use of own concept paper is not allowed.
- Use only indelible ink use of pencils and red ink is prohibited.
- Other aids than that mentioned above is strictly forbidden. This includes any type of communication to other people.

#### Duration of the examination

120 minutes

#### **Examination documents**

The examination comprises 30 pages (including title page). Answers may be given in English or German. A mix of language within a single (sub)-task is not allowed. In your solution mark clearly which part of the task you are solving. Do not write on the backside of the solution sheets. If additional paper is needed ask the examination supervisor. It may not be possible to finish all tasks within the duration of the examination. This will be accounted for within the grading of the exam.

You will not be allowed to hand in your examination and leave the lecture hall in the last 30 minutes of the examination.

At the end of the examination: Stay at your seat and put all sheets into the envelope. Only sheets in the envelope will be corrected. We will collect the examination.

		Page	~ Pts [%]	Points
Task 1	Error Protection	2	17%	18
Task 2	Media Access	6	14%	16
Task 3	Synchronization	11	13%	15
Task 4	Data Transmission	15	8%	10
Task 5	Physics	17	15%	17
Task 6	Practical Aspects of Communication Systems	21	15%	17
Task 7	Networks	26	18%	20
				Σ 112

# Task 1 Error Protection

## Task 1.1 Error Detection

A) Name two methods for error detection within a communication protocol based on redundancy.

Parity Check, Block Check, XOR Block Check, Hamming Code

B) To which extend can Parity Checking be used to detect burst errors? Justify your answer.

Parity checking can detect all burst errors with an odd number of errors

## Task 1.2 CRC-Calculation

The bitstream 10110011 shall be coded and transmitted using the generator polynomial  $g(x)=x^6+x^5+x^3+x^2+1$ .

- A) Give the bitstring for the given generator polynomial
   Bitstring for generator polynomial: 1101101
   Point is also given if bitstring is given correct in B) or C)
- B) Determine the bitstream as it is being transmitted.

Bitstring for generator polynomial: 110110

10110011000000

01001

0000100000

0101010

<u>1101101</u> 01101110

1101101

0000011

Bit stream as it is transmitted: 10110011000011

01

/	· . /	·	$\searrow$					
Degree of generator	` <b>D⊘</b> ł∕v	nomial	= 6. h	ence si	x zeros	need to	be	appended
	17-17		-,					

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Remainder: 000011

1pt:

2pts:

**1pt**:

bitstream appended with six 0's

1pt if single calculation error 0pt more than 1 calc error

correct bitstream that is being

Opts if systematic error

transmitted

calculation correct (remainder = 0)

18

1

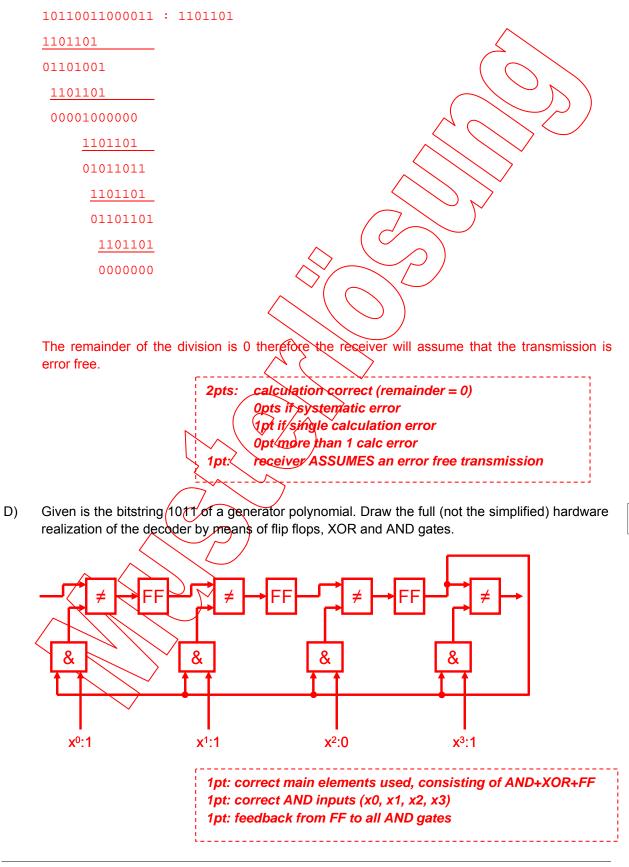
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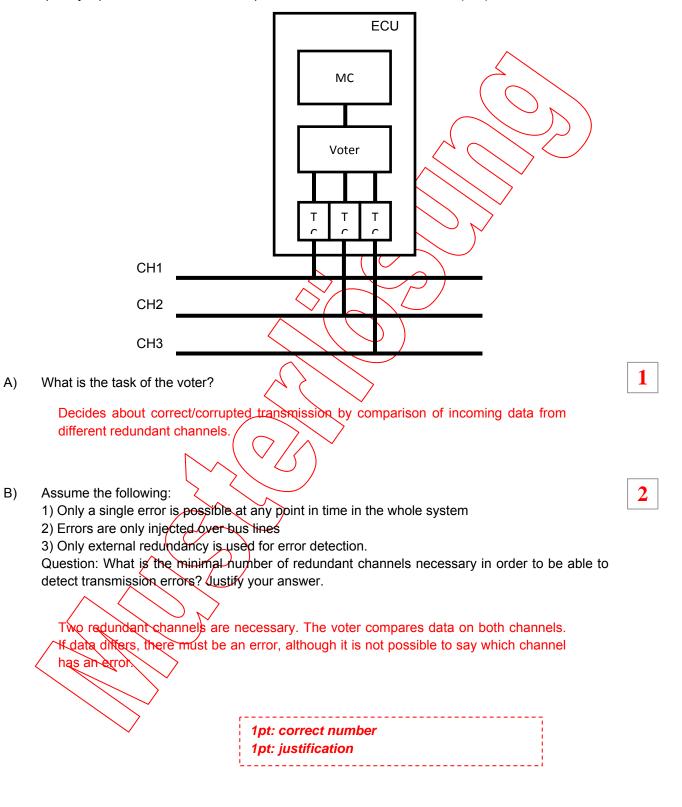
C) With a transmission system that uses CRC for error protection, a sender transmits the following bitstream: 10110011000011.

Carry out the CRC error detection scheme of the receiver, assuming that the generator polynomial  $g(x)=x^6+x^5+x^3+x^2+1$  has been used. What does the receiver conclude from the result?



## Task 1.3 External Redundancy

The figure below illustrates a setup of a bus system using multiple parallel channels (CH1, CH2, CH3). Data that is received by an electronic control unit (ECU) is read off the bus via transceivers (TC). Subsequently it passes a voter before it is processed within a microcontroller (MC).



C) Assume the same preconditions as in B). How many redundant channels are necessary in order to be able to correct transmission errors? Justify your answer.

Three redundant channels are necessary. If all three channels deliver the same value, either all three are working properly (most likely) or all three have failed in same manner (most unlikely). If two have the same value, most likely the channel that differs has failed and we can rely on the value data delivered by the two identical channels -> Hence correction is possible.

1pt: correct number 1pt: justification

#### Task 2 Media Access **Task 2.1** Multiple use of media A) For transmitting data of multiple nodes at the same time different access schemes exist. Name 1 two access schemes other than CDMA. CDMA = Code Division Multiple Access FDMA = Frequency Division Multiple Access TDMA = Time Division Multiple Access SDMA = Space Division Multiple Access 0.5 points per scheme 1 B) Give the Walsh-functions for the transmission of four nodes at the same time Function 1 +1+1 Point only given for correct table CAREFUL: Order can be Function 2 +1different Also correct substitution: Function 3 +1 +1 -1 -1 +1 = 0-1 = 1 Or vice versa Function 4 +1 +1 Table 2.1: Walsh-functions

For a CDMA access scheme the first three chips of the Walsh-functions from the subtask above got lost. Sending the bits listed in table 2.2 the signal **0 4 0 0** can be measured on the medium.

			Walsh-function			send bit
	A	74	1	+1	-1	1
	В		+1	+1	+1	0
	C	+1	+1	-1	-1	0
	D	+1	-1	-1	+1	1
Table 2.2: CDMA scheme using Walsh-functions						

C)					vith the correct missing chips. Justify your answer by giving all your ur complete reasoning.	
	+1	+1	+1	+1		_
	0	4	0	0		
	0	+4	0	0	0 was send	
					Last bit of Walsh is +1 and 0 was send, therefore has to be node B	
	+1	-1	+1	-1		
	0	4	0	0		
	0	-4	0	0	1 was send	
					Last bit of Walsh is - 1 and 1 was send, therefore has to be node A	
	+1	+1	-1	-1		
	0	4	0	0		
	0	+4	0	0	0 was send ast bit of Walso is -1 and 0 was send, therefore has to be node C	
	+1	-1	-1	+1		
	0	4	0		$\tilde{\langle}$	
	0 <	-4	$\langle \rangle$	0	1 was send	
	$\wedge$	$\sum$		$\searrow$	Last bit of Walsh is +1 and <u>1 was send, therefore has to be node D</u>	
	$\sim$			$\searrow$	<sup>7</sup> 1pt: correct function assignment	
2 <sup>nd</sup> se	olution:		$\sum$	$\checkmark$	Points when using calculations: 2pts: for correct calculations of correlation)	
					has to be +4. ust be inverted.	
Assu	ming th	at se	nding	'1' me	eans inversion of <b>Points when using reasoning:</b>	
					2 and 4 can be assigned3pts for correct reasoningrsa). Because the1pt for ansatz (2 <sup>nd</sup> chip is resulting in +4)	
	hip of f Ined to			-1 it	nas to be 1pt for ansatz (inverting functions)	
<u>assiy</u>		iuncu	UT D.		1pt for assigning 1 and 0 (deciding what function is inverted)	

3

D) What are the general advantages and disadvantages of CDMA? (list one advantage and one disadvantage)

0.5 points for each correct answer

Pro:

More robust against narrowband disturbances because signal energy is distributed on a broader spectrum

Lower sensitivity against interferences from other nodes

Better protection against eavesdropping As the signal looks like white noise, it is impossible to detect when a transmission is going on

Eavesdropper has to know the used pseudo random sequence in order to be able to reconstruct the send data

Contra:

More effort required an sender and receiver side

Requires channel with larger bandwidth

### Task 2.2 Arbitration

A) For the arbitration of four nodes a Tap-line model is used. Draw a Tap-line model with four nodes. Label all used signals. Give a short explanation (one sentence) of each of the different types of signals used in this scheme. Mark the arbiter in your scheme and briefly explain its basic function (ca. 3 sentences).

Request: If node wants to send, it signalize it with dominant level on Request line

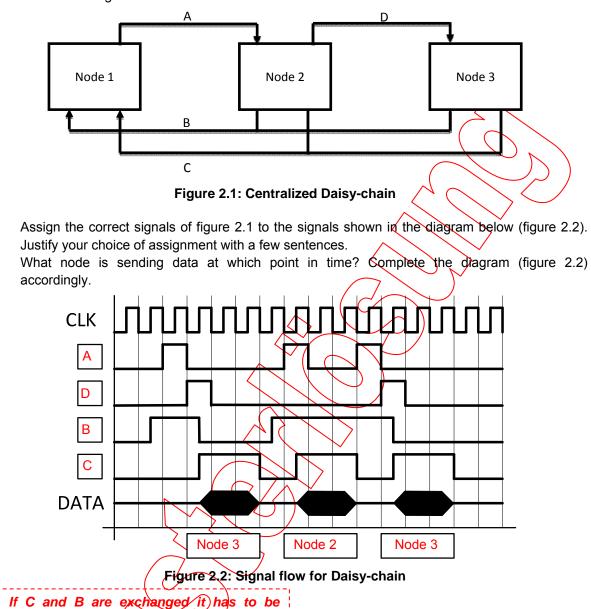
Grant: gives access of transmission/line to one exclusive node

0.5 points for each explanation

1 point for correct description

Node signalizes his request on the request line. The arbiter gives grant to one exclusive node, regarding the priority initialized in the arbiter.

A system using centralized daisy-chaining is shown in figure 2.1. An exemplary arbitration cycle of the system is shown in figure 2.2.



If C and B are exchanged it) has to exchanged at explanation as well.

Lowest signal has to be C/B because it is the busy line, which is high while data is transmitted.

Second to last signal has to be B/C, because it is a request line which triggers the token/grant and gets low while sending.

A and D are both grant or token signals.

First has to be A, because it is a token/grant and it starts first.

Second has to be D, because it is a grant and it always starts after A.

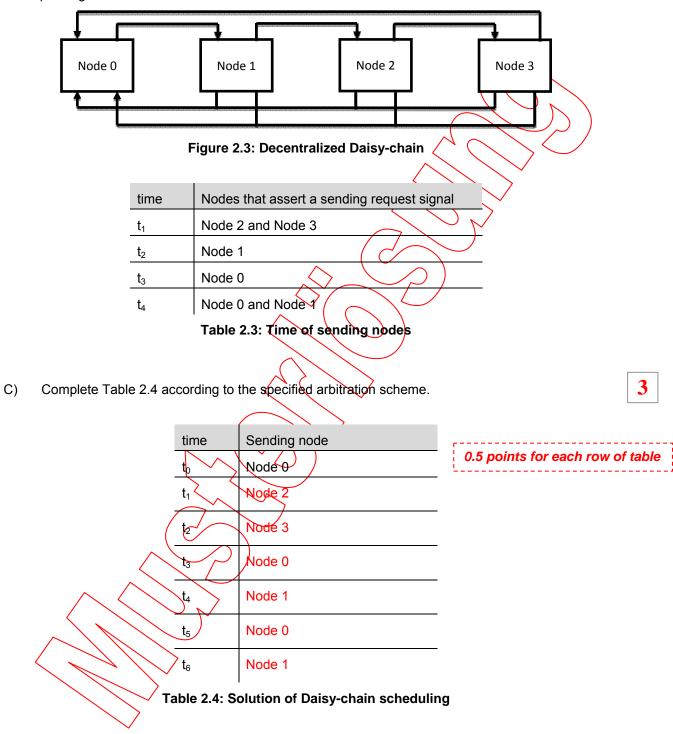
## 2 Points for left labels

+0.5 points for each correct description and label in Figure 2.2

1 Point for correct sending nodes +0.5 points if only two are right

B)

In the decentralized Daisy-chain shown in figure 2.3 a scheduling should be done. The different nodes will set a request at the times given in table 2.3. Only after successful transmission the nodes will remove their request. The sending of the data always needs exactly one time step. This includes token passing and the time needed for the arbitration.



# Task 3 Synchronization

# Task 3.1 Synchronization methods

A) Characterize the synchronization methods in the table below with respect to the given transmission types.

		-	-	<u></u>
Transmission methods	Parallel	Serial	Synchronous	Asynchronous
Shared/dedicated clock line	x		x	
Start-stop mode		x		
Suitable line code		x	×	
Handshake mode	x			×
Scrambler		~*	x	
	~ ~			

min.

-0.5P)for each false/missing X

B) Using the table below, compare the given methods with respect to the synchronization characteristics/properties and cost. Only one example per cell is necessary.

Method	Advantages	Disadvantage
Dedicated clock line	Easy to determine the bit intervals     No data signal modification needed	<ul> <li>More signal line =&gt; expensive</li> <li>Parallel transmission =&gt; skew</li> </ul>
Suitable tine code	<ul> <li>Less signal lines =&gt; less cost</li> <li>Timing recovery</li> </ul>	<ul> <li>higher bandwidth needed =&gt; expensive</li> </ul>
Scrambler	<ul> <li>Less signal lines =&gt; less cost</li> <li>Timing recovery</li> </ul>	<ul> <li>Additional logics needed for inserting and extracting scrambler bits =&gt; expensive</li> </ul>

+0.5P for each correct box

## Task 3.2 Hand-shaking partners

A communication system is given in **Fehler! Verweisquelle konnte nicht gefunden werden.** The sender's clock frequency is 1 MHz, the receiver's is 200 kHz. Both partners work synchronously to their own clock signal and try their best to communicate as fast as possible. They apply a half-duplex hand-shake protocol corresponding to Figure 3.1 for the high-level synchronization.

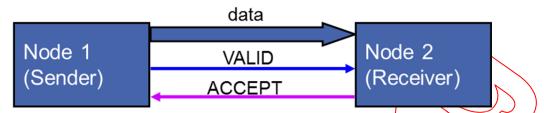
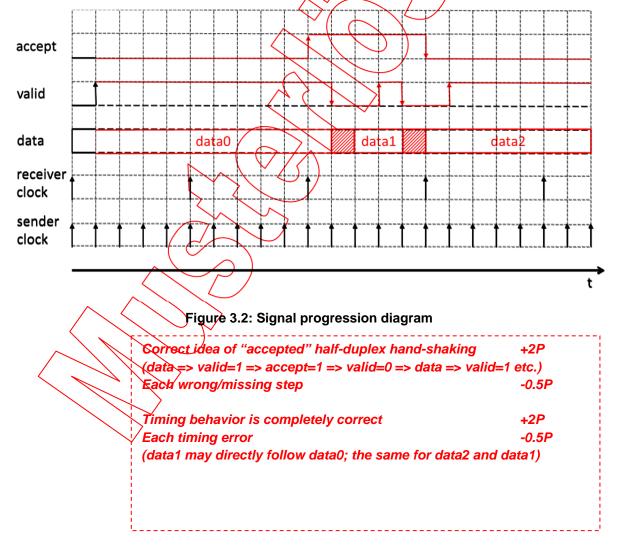


Figure 3.1: Communication system applying half-duplex hand-shake procedure

A) In Figure 3.2, the sensitive clock edges of the sender and the receiver as well as the signals' values for the first sender's clock period are shown. In order to avoid violations of setup and hold times, the data is put onto the bus and one clock cycle later the *valid* signal is set to '1' by the sender. The receiver will also set the *accept* signal one clock cycle after having received the data. Fill in the progression of all signal thes until the end of the time scale.



1

B) Is this kind of synchronization free from error in this specific case? Justify your answer.

No. One can see that data1 got lost because it was put on to the data lines and removed by sender while receiver is still busy processing data0.

Recognize that data1 is lost+0.5PCorrect reasoning for the data lost+0.5P

1pt; One correct solution

- C) Propose a better solution for this communication scenario.
  - 1. Clock divider in sender so that both are only clocked at 200 kHz.
  - 2. Apply a full-duplex hand-shaking protocol.
  - 3. Clock down the sender: to avoid this loss of data, the following equation must be satisfied:

### 3Tr ≤ 2Tr + 4Ts

### ⇔ Tr≤4Ts

The receiver's clock frequency of must be more than or equal to one-fourth of the sender's one.

## Task 3.3 Against noise with tied hands

You are to design a data communication system with a data rate of 25 Mbps. However, the given communication channel is noisy. Statistics shows that during one bit interval, one noise pulse, which causes the signal to be misinterpreted (bit flipping), may occur with a maximum duration of 15 ns. Due to resource constraints, neither are you allowed to switch to another channel, nor is it possible to modify the sender, which does not integrate any mechanism for error correction. What can you do in order to make your system works correctly (statistically seen)? Rationalize your solution.

Oversampling with minimum factor k (due to resource constraints) is needed => k samples available. Data rate r = 25 Mbps => Bit interval I = 1 / (25x10^6) = 40 ns

If k=1 => the only sample point may be overlapped => bit flipping may occur. If k=2 => distance between sample points is d = 40/2 = 20 ns, noise pulses may only overlap 1 of them. However, there is no way to distinguish which sample point holds the right value. If k=3 => d = 13,3 ns, noise pulses may overlap 2 of 3 sample points => bit flipping may occur. If k=4 => d = 10 ns, noise pulses may overlap 2 of 4 sample points => same situation as if k=2 If k=5 => d = 8 ns, noise pulses may overlap max. 2 of 5 sample points => the bit can be recognized correctly using majority voting.

Result: oversampling with factor 5 is needed/

#### Alternative solution: low-pass filter!/

<u> </u>	<b>\</b>		
	Correct method: oversampling	+0.5P	
Ι	Minimal value of k needed	+0.5P	j
	Correct bit interval	+0.5P	1
	Correct value of k	+0.5P	j
	Reasoning for value of k	+1P	
	-		

# Task 4 Data Transmission

## Task 4.1Transmission Rate

The CSMA/CD media access control as used in Ethernet includes a collision detection method. Thereby, a sender transmits a jam signal if it detects another signal while transmitting a frame. The jam signal must propagate to all receivers before the transmission ends. This introduces a minimum frame size for Ethernet that is coupled to the maximum wire length. The minimum frame transmission time must be higher than the time required passing a maximum length wire twice.

A) Calculate the minimum frame size in bytes of an Ethernet network running at 100 Mbit/s over a coaxial cable with the following parameters?

Transmission rate: 100 Mbit/s

Maximum cable length: 500 m

Propagation speed of coaxial cable: 200000 km/s

Minimum frame transmission time: 1 km / 200060 km/s =  $1/2 \times 10^{1/2}$  = 5 \* 10^-6 s

Minimum frame size:  $5*10^{-6}$  s \*  $1*10^{-8}$  Bits/s  $\in 500$  Bits

- = [500/8] Byte = 62,5 Byte
- ➔ When sending full bytes: 63 Byte
- B) Calculate the minimum efficiency of the transmission system when sending only dataframes with minimum frame size and an overhead of 100 bits per frame are required.

(500 – 100 bits) / 500 bits = 80%

1 pts: correct result 0,5pts: 500/(500+100)=83,333...

2 pts: correct result

Else 1pt: correct Ansatz

C) The maximum frame size for ethernet is 1500 bytes. Would you use CSMA/CD together with 10 Gbit Ethernet? Please explain your answer.

No, the minimum frame size for 10 Gbit Ethernet is 50,000 bits and this is larger than the maximum frame size.

1 pt: correct answer 1 pt: explanation

1

3

### Task 4.2 Shannon Limit

A) What is the Shannon Limit?

The Shannon Limit tells the maximum rate at which information can be transmitted over a communications channel of a specified bandwidth in the presence of noise.

B) Calculate the Shannon Limit for a channel with 1000 Hz bandwidth and a S/N of 127. Give the result in bits/second.

CS = fmax<sup>\*</sup> log2(1+S/N) = 1000 Hz<sup>\*</sup> log2(128) bit = 7000 bit/s = 7 kbit/s

C) Calculate the required signal-to-noise ratio (SNR) in dB for a channel with 1 MHz bandwidth and a transfer rate of 3 Mbit/s.

#### $CS = fmax \cdot * log2(1+S/N)$

- $\Rightarrow$  Log2(1+S/N) = CS / fma
- ⇒ S/N = 2^ (CS / fmax) 1
- $\Rightarrow$  S/N = 2 ^ (3 Mbit/s / 1 + Mbz) 1 = 2^3 1 = 7
- ⇒ SNR = 10 log(7) = 8,95 db

 pt: Ansatz (take formula and reorder it to S/N)
 pt: dB conversion (method)
 pts: calculation of S/N correct
 pts: dB conversion correct

1 pt: correct answell

1/pt:)everything correct

# Task 5 Physics

# Task 5.1General questions

A) What are parameters of a periodic signal? Name 2 parameters.

Period T or Frequency 1/T Amplitude S(t) Phase  $\phi$ 

0.5 pt per parameter (1 pt max)

B) Several modulation techniques were discussed in the lecture. Name 3 of them and describe 3 shortly their functionality.

Amplitude Modulation, Amplitude Shift Keying Linear change in amplitude in radio broadcasting.

Frequency Modulation, Frequency Shift Keying

Linear change in frequency in radio broadcasting

Phase Modulation, Phase Shift Keying

Phase jumps in digital systems.

#### **Quadrature Amplitude Modulation**

Combination of ASK and PSK.

#### **Pulse Modulation**

Varying a digital signal to represent an analog value.

1 pt for naming and description of 1 modulation technique 0.5 pt for naming of 1 modulation technique

C) What does cut off frequency mean?

This is the frequency at which the signal amplitude has dropped by 3 dB compared to the output value.

1 pt for correct answer 0.5 pt not possible

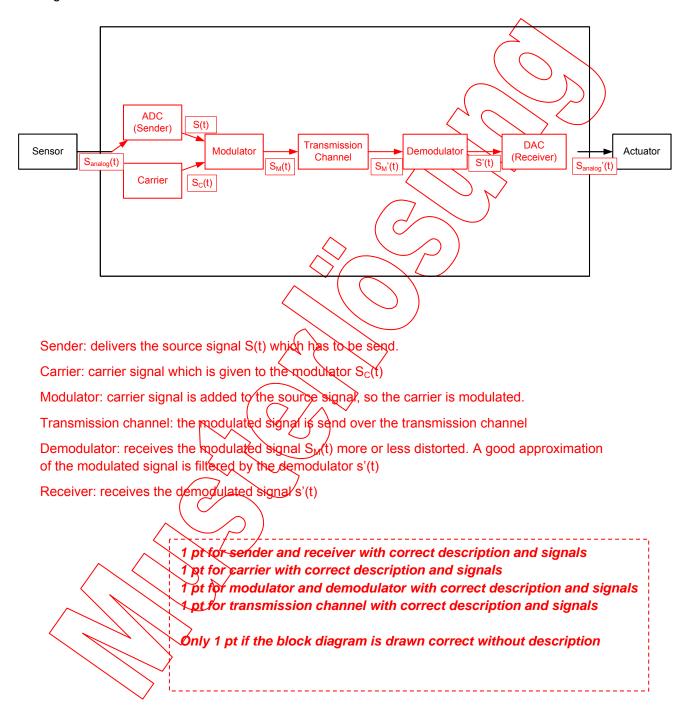
# Task 5.2 DA/AD-Conversion

A) What is the minimum frequency an analog signal has to be sampled with in order to allow perfect reconstruction of the analog signal? Give the equation and the name of the theorem.

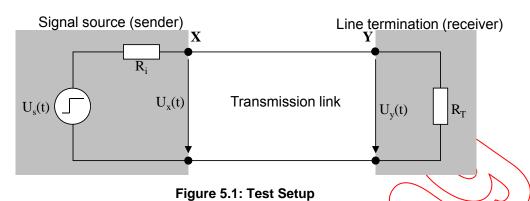
The sample & hold element holds the analog signal constant during the AD conversion.          1       1         1       1         0.5       pt nor correct answer         1       N sample points per bit (n-times oversampling)         Majority voting among the samples → value that appears most is taken         Advantage: Filtering of short distortions is possible         Analog oversampling:         Multiple sampling of the signal <tr< th=""><th></th><th><math>f_{sample} \ge 2 * f_{max}</math></th><th></th></tr<>		$f_{sample} \ge 2 * f_{max}$	
<ul> <li>3) Why and when is a sample &amp; hold element necessary?</li> <li>1</li> <li>The sample &amp; hold element holds the analog signal constant during the AD conversion.</li> <li>1 pt for correct answer</li> <li>0.5 pt not possible</li> <li>C) Oversampling: Explain what oversampling means and name the advantages.</li> <li>Digital oversampling:</li> <li>N sample points per bit (n-times oversampling)</li> <li>Majority voting among the samples → value that appears most is taken</li> <li>Advantage: Filtering of short distortions is possible</li> <li>Analog oversampling:</li> <li>Multiple sampling of the signal</li> <li>Sampling steps has to be in the specified range.</li> <li>Advantage: Compensation of failures</li> </ul>	Nyquist-Shannon-Sampling The	orem	
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The sample & hold element holds the analog signal constant during the AD conversion.          1 pt for correct answer       0.5 pt nor possible         C)       Oversampling: Explain what oversampling means and name the advantages.         Digital oversampling:       N sample points per bit (n-times oversampling)         Majority voting among the samples → value that appears most is taken         Advantage: Filtering of short distortions is possible         Analog oversampling:         Multiple sampling of the signal         Sampling steps has to be in the specified range.         Advantage: Compensation of failures		1 pt for correct theorem and equation	
1 pt for correct answer       0.5 pt not possible         C)       Oversampling: Explain what oversampling means and name the advantages.         Digital oversampling:         N sample points per bit (n-times oversampling)         Majority voting among the samples -> value that appears most is taken         Advantage: Filtering of short distortions is possible         Analog oversampling:         Multiple sampling of the signal         Sampling steps has to be in the specified range.         Advantage: Compensation of failures	B) Why and when is a sample &	hold element necessary?	1
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0.5 pt not possible         C)       Oversampling: Explain what oversampling means and name the advantages.         Digital oversampling:         N sample points per bit (n-times oversampling)         Majority voting among the samples → value that appears most is taken         Advantage: Filtering of short distortions is possible         Analog oversampling:         Multiple sampling of the signat         Sampling steps has to be in the specified range.         Advantage: Compensation of failures			
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Advantage: Compensation of failures	Multiple sampling of the signal	$\sim$	
	Sampling steps has to be in the s	pecified range.	
	Advantage: Compensation of failt	ures	
<b>A A A A A A A A A A</b>		1 pt explained digital or analog oversampling with	h
advantages		advantages	
0.5 pt for advantage or explained oversampling		0.5 pt for advantage or explained oversampling	

## Task 5.3 Modulation

An analog signal is send from a sensor to an analog actuator. It has to be modulated with frequency shift keying. Draw in a block diagram the complete transmission chain of the signal and name all necessary blocks with their corresponding signals. Give a short description of the task of each block and signal.

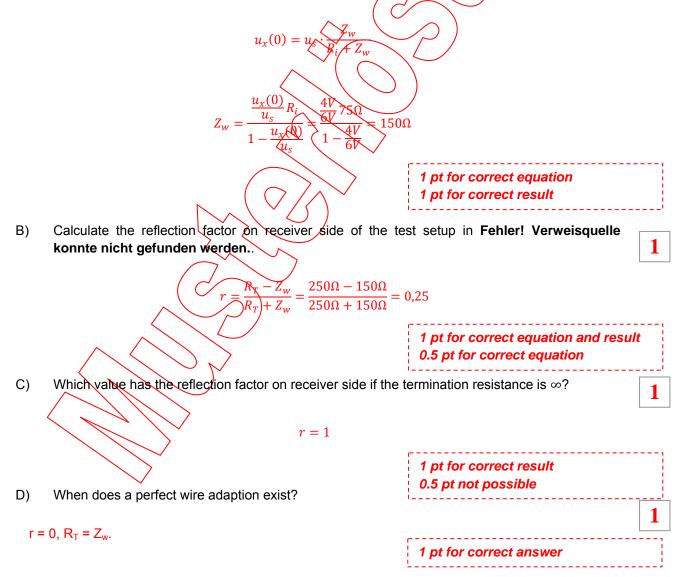


### Task 5.4 Reflection on wires



A) In Figure 5.1 an assembly is considered, consisting of a voltage source with an internal resistance  $R_i = 75\Omega$  as sender and a receiver with  $R_T = 250\Omega$ . The DC resistance of the line is zero. Calculate the value of the wave resistance at the time of t=0.

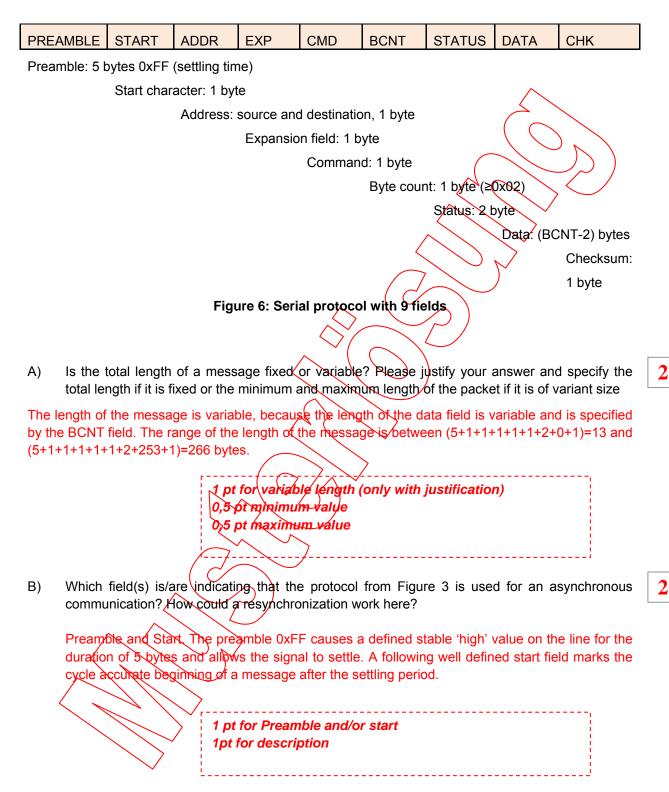
At the time t=0 the voltage  $U_s$  of the sender changes from 0V to 6V and is constant afterwards. The runtime of a wave on the wire is t<sub>d</sub>. The voltage  $u_x$  at the time of t=0 is  $U_x(0) = 4V$ .



# **Practical Aspects of Communication Systems** Task 6 **Task 6.1 General Questions** A) Which factors are constraining the throughput of parallel busses? Name 2 factors. 1 Bus width Skew (routing) Clock frequency (critical path, parasitics, fanout, input capacitances, ...) Attenuation What is the difference between synchronous and asynchronous communication? B) 1 In synchronous communication a clock signal is transmitted along with the data. No synchronization logic is needed in the receiver. In asynchronous communication there is no dedicated clock signal. The received data needs to be synchronized and in case of variable baud rate, a clock recovery needs to be done on the receiver side. C) How many layers does the TCP/IP internet reference model have? four Explain the difference between a LAN switch and a router. D) A LAN switch is setting up point to point connections (no broadcast compared to HUBs) 1 between the participants in a local area network. A router is routing network packets among several networks. Router works on Level 3 and switches on Level1 or Level 2.

## Task 6.2 Serial Protocol

In Figure 6 a serial protocol is given.

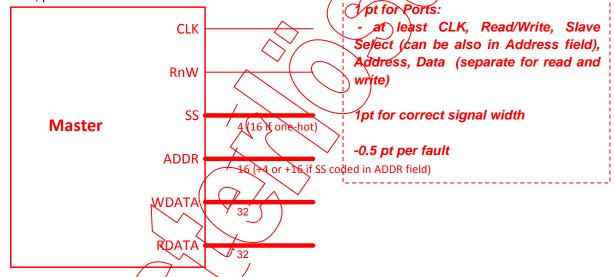


### Task 6.3Development of a custom bus

For a specialized system-on-chip a synchronous on-chip bus for communication between a master module and several slave modules needs to be developed.

The requirements are:

- Clock synchronous
- Parallel bus
- No tri-state drivers
- Separate bus lines for read data and write data
- Width of data busses is 32 bit
- Address range from 0x0000 to 0xFFFF in each slave
- Maximum 16 slave modules possible
- A) Please draw a "black box" of the master module, containing all necessary ports. In case of bus lines, please mark the bus widths.



B) Is there a difference in the slave modules concerning the signals or the widths? Justify your answer?

Yes, the slave module needs only one select signal ...

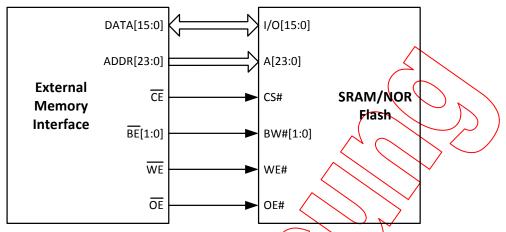
or (in case SS is integrated in ADDR field):

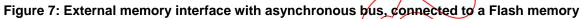
Yes, the address width is smaller since only the bit indicating the respective slave is needed...

No, the slave select is decoded inside the slave module ...

## Task 6.4 Timing diagram of asynchronous bus

In Figure 7 an asynchronous parallel bus for interfacing e.g. NOR Flash memories is given. A description of the interface signals is given in Table 4.





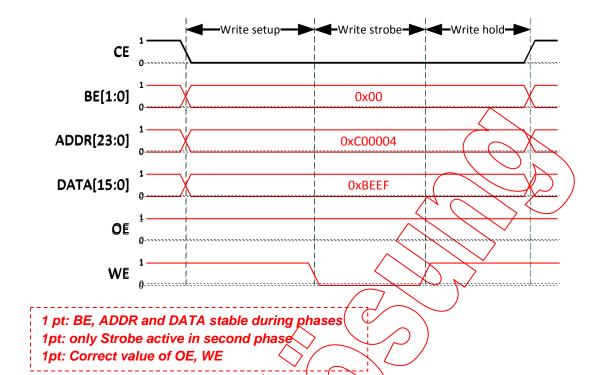
Port	Description
DATA [15:0]	Data I/O pins. 16/8-bit bidirectional data path for I/O.
ADDR [23:0]	External address outputs
CE[3:0]	Chip select for CE space. Active-low chip select for memory spaces 0 to 3.
BE[1:0]	Active-low byte enables (Upper and lower). Individual bytes or half-words can be selected: "00" = data on DATA[15:0], "01" = data on DATA[15:8], "10" = data on DATA[7:0]
OE	Active-low output enable. Low during read access period.
WE	Active-low write enable. Low during write transfer strobe period.

### Table 4: Description of the interface signals

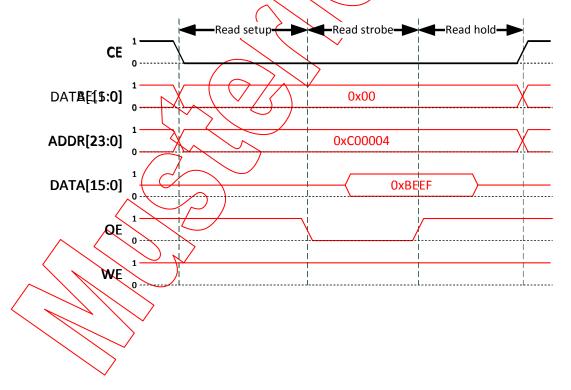
Since the bus is asynchronous, read and write accesses are taking place in three phases:

- 1. **Setup:** Phase between the beginning of a memory cycle (chip select low, address valid, byte enable valid, write data valid) and the activation of read or write strobe.
- 2. Strobe: Phase between the activation and deactivation of the read (OE) or write (WE) strobe.
- 3. **Held:** Phase between the deactivation of the read or write strobe and the end of the cycle (which may be either an address change or the deactivation of the chip select signal.

A) Please complete the signal changes in the timing diagram below for a write access of data 0xBEEF at address 0xC00004! Consider the above named phases!



B) Please complete the signal changes in the timing diagram below for a read access at address 0xC00004! Consider the above named phases!



# Task 7 Networks

# Task 7.1 XY-Routing

Figure 7.1 shows a 4x4 meshed Network-on-Chip with bidirectional links for packet based communication.

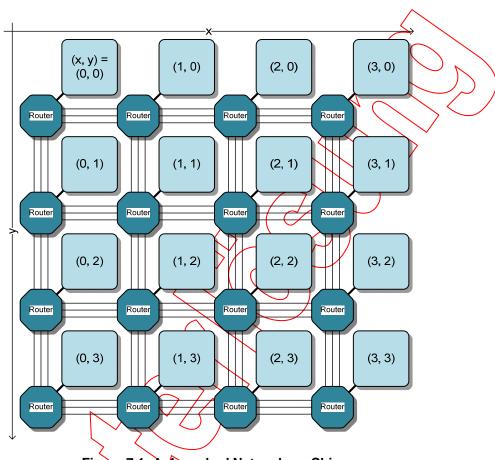


Figure 7.1: 4x4 meshed Network-on-Chip

For the following it is assumed, that XY-routing is used.

A) Which routers are passed by a packet sent from (x, y) = (1, 1) to (2, 3). Please provide the coordinates of the passed router in the order given by the transmission process.

+ 0.5 Pt. per 2 node



(1, 1), (2, 1), (2, 2), (2, 3)

1

Now, there is an error on the NoC, blocking the link between router (1, 1) and (0, 1).

B) Is router (3, 1) affected by this error? If so, which are the destination nodes to which router (3, 1) cannot communicate anymore (give the coordinates)?

+ 0.5 Pt. per 2 node

#### (0, 0), (0, 1), (0, 2), (0, 3)

C) XY-Routing enables deadlock-freedom by avoiding turns. Which turns are these?
 South-West, South-East, North-West, North-East
 Turns from column into a row are prohibited
 180° turns
 1pt: correct explanation or complete list of correct turns

D) Comparing centralized and distributed routing. Give one advantage for each strategy <u>Centralized Routing: global optimization possible</u>,

reduced complexity in each router

Distributed Routing: scalability due to no communication overhead,

Fast reaction on changing routing requirements

+ 1 Pt. per valid advantage

E) Besides centralized and distributed routing there exist other routing methods. Give two of them.

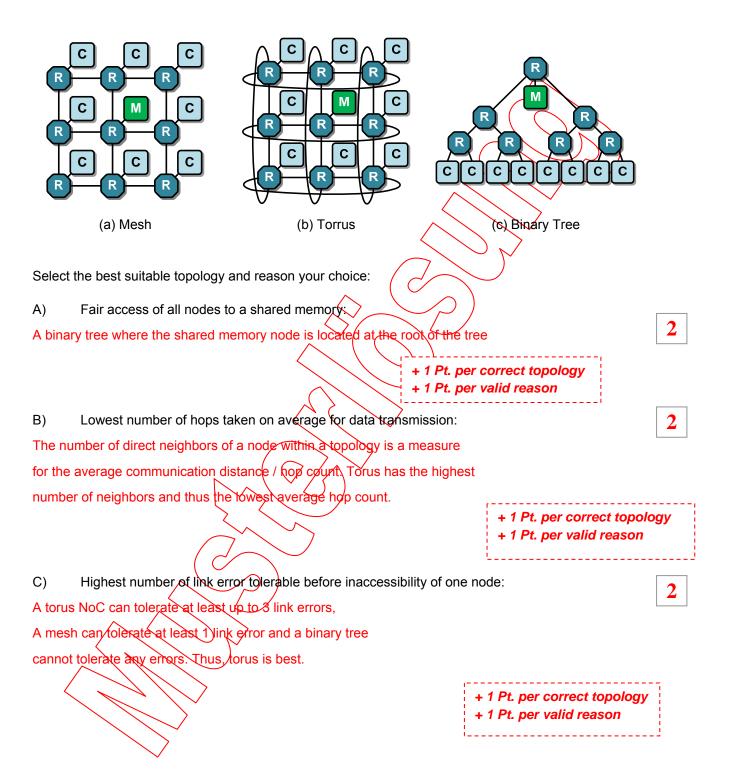
Source Routing, Static Routing, Adaptive/Dynamic Routing,

Unicast Routing, Multicast Routing, Broadcast (e.g. Flooding)

+ 0.5 Pt. per routing method

## Task 7.2 Topologies

In the following three different topologies: (a) Mesh, (b) Torus and (c) Binary Tree will be investigated under different constraints. All links can be assumed to be bi-directional.



0.5 Pt. if correct

3

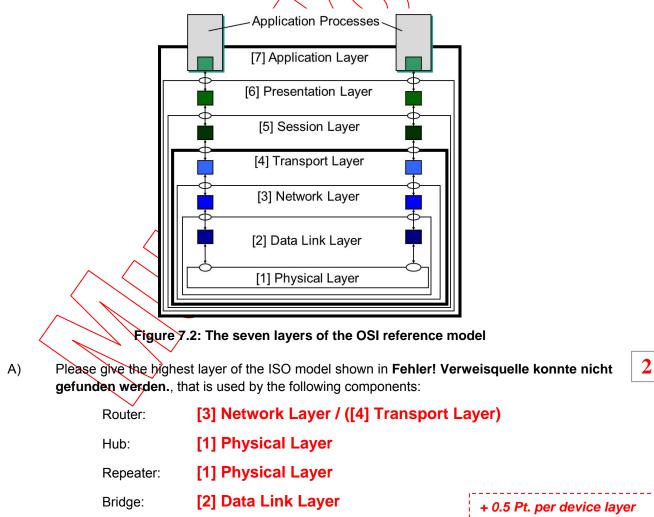
## Task 7.3 Circuit- and Packet-Switching

In the following different requirements are given. Please select the switching method that fits best to the given requirement.

Requirement	Circuits-Switching	Packet-Switching
Low latency (transmission of 1 byte of data, no connections established)		X
Low latency (transmission of 1 byte of data, connections established)	X	
Real time guarantees	X	
Energy Consumption	X	
Flexibility	<	
Short and non-frequent data transfers		$\times$

# Task 7.4 OSI Reference Model

Figure 7.2 shows the 7 layers and the layer names of the OSI reference model



The following latencies for data processing within each layer of the OSI reference model can be assumed for all devices in the following:

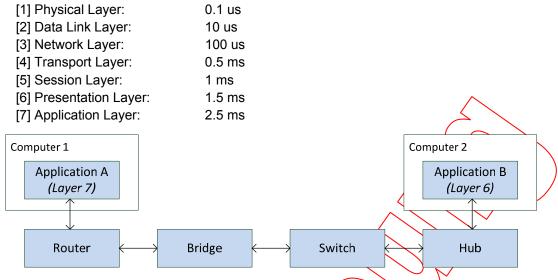


Figure 7.3: Data transmission scenario

In Figure 7.3 the communication between Application A and Application B is shown. The payload size can be assumed as small. Thus, data transmission latency is assumed to be independent from the payload size. Links between devices are short and transmission delay can be assumed to be zero. Communciation between devices is done on the physical layer.

B) Please calculate the communication latency of each device on the communication path between Application A and Application B shown in Figure 7.3.

Computer 1:

Layer[7] + Layer[6] + Layer[5] + Layer[4] + Layer[3] + Layer[2] + Layer[1] = 5,6101 ms

Router:

Layer[1] = 220,2 us 2 \* Layer[3] + 2 \* Layer[2] + 2 \*

20

Bridge:

2 \* Layer[2] + 2 \* Layer[1]

```
Switch:
```

```
2 * Laver[2] + 2 * Laver[1] = 20,2 us
```

+ 0.5 Pt. per device latency

Hub:

```
2 * Layer[1] = 0.2 us
```

```
Computer 2:
```

```
Layer[6] + Layer[5] + Layer[4] + Layer[3] + Layer[2] + Layer[1] = 3,1101 ms
```