

Real-Time Programming

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Chapter 1

Real-Time Systems

Digital, computer-based real-time systems became widespread rapidly with the availability of microprocessors getting more powerful and cheaper from year to year.

1.1 Introduction

Real-Time systems, as used in the wide area of automation, have 2 basic requirements,

1. The **logical correctness** of the systems output as a response to its inputs, and
2. The **timeliness** of the outputs available.

For example, this is obvious for an airbag control, for which a correct decision for ignition is required at the right millisecond, otherwise the resulting behaviour can be harmful !

Since there is a "digital revolution" the past few decades, a strong trend to applications with microprocessors and with microcontrollers particularly can be observed.

This means, that classical solutions for **real-time systems** (e.g. analog controller circuits) are replaced by **software applications** running on a microprocessor or a microcontroller.

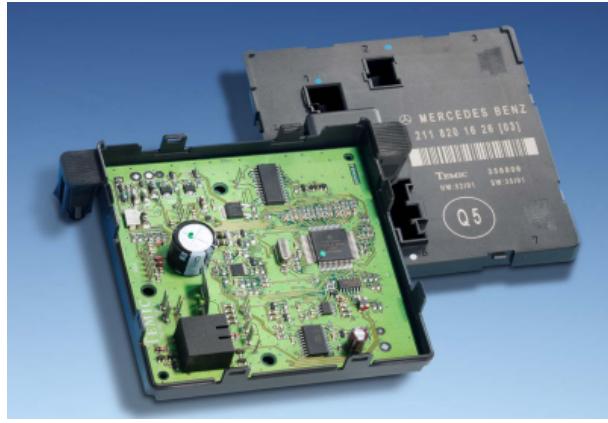
Thus, an important part of **a real-time system is software**. This type of software is much different from commonly used software known from PC applications like office programs, or internet browsers, since real-time requirements have to be met.

In the area of real-time software development, the classical C programming language is widely used, due to its outstanding maturity and flexibility, especially with industrial PC applications, or with microcontroller applications.

Furthermore there is a need for PLC controls (de: SPS) in industrial automation.

Examples for Real-Time Systems

- Examples of the use of real-time systems are found in the following areas:
 1. Production
 2. Aerospace
 3. Automotive Electronics



(a) Automotive Door Control Unit with Anti-Squeeze



(b) Undistorted image



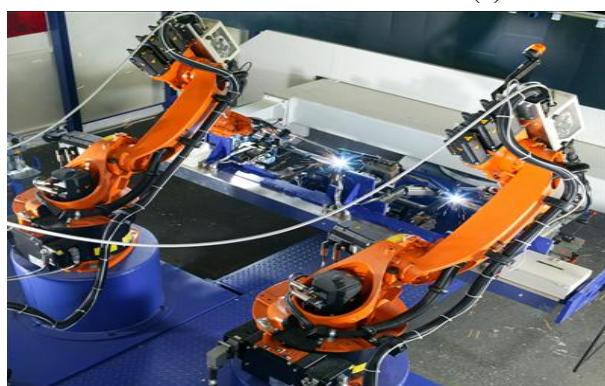
(c) Car with > 100 ECUs



(d) Pacemaker



(e) A319 Cockpit



(f) wikipedia.de: Arc welding of metal parts with industrial robots (KUKA). IPC, PLC

Figure 1.1.1: Comparison of results of video 1 (checkerboard000)

4. Medicine
5. Military Technology,
6. e-Banking,
7. e-Trading,
8. Telecommunications,
9. Network Management,
10. Power Generation/Management,
11. Navigation

The need for Software(mainly C-Code)in all areas of application grows rapidly.

Moore's Law: "die Anzahl der Transistoren pro Chipfläche verdoppelt sich alle 2 Jahre"

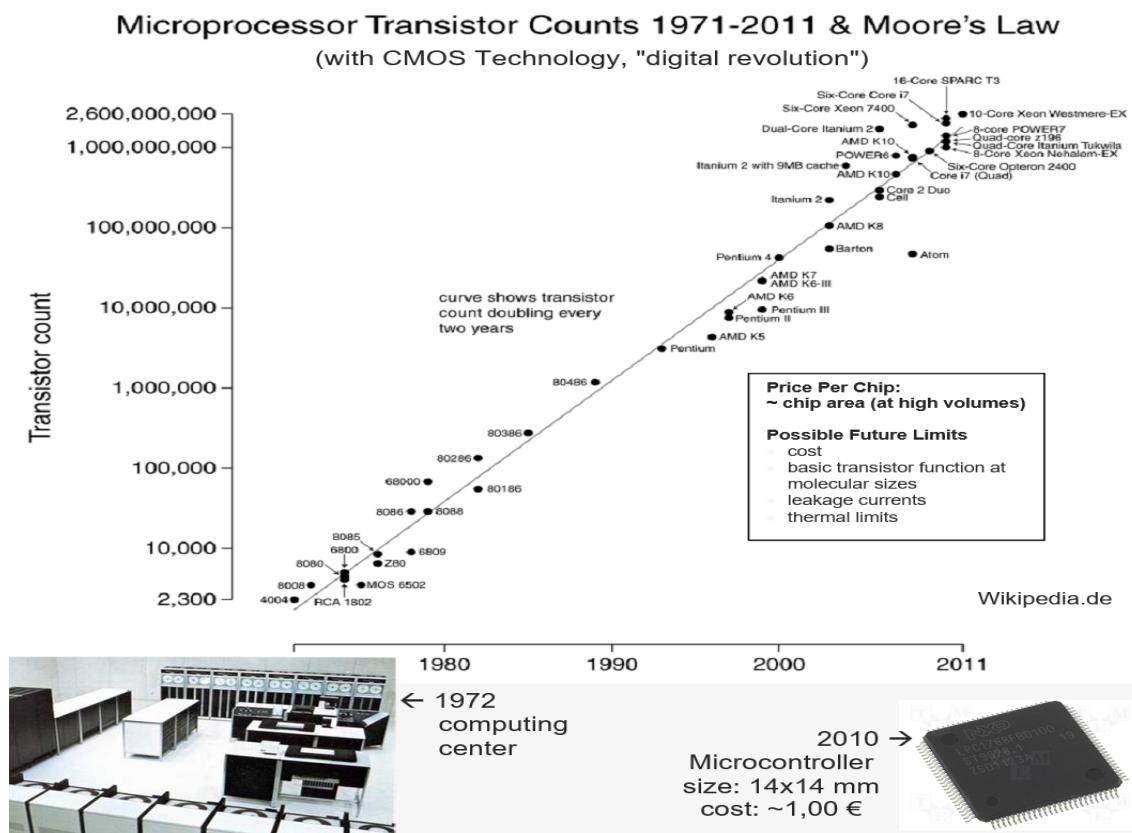
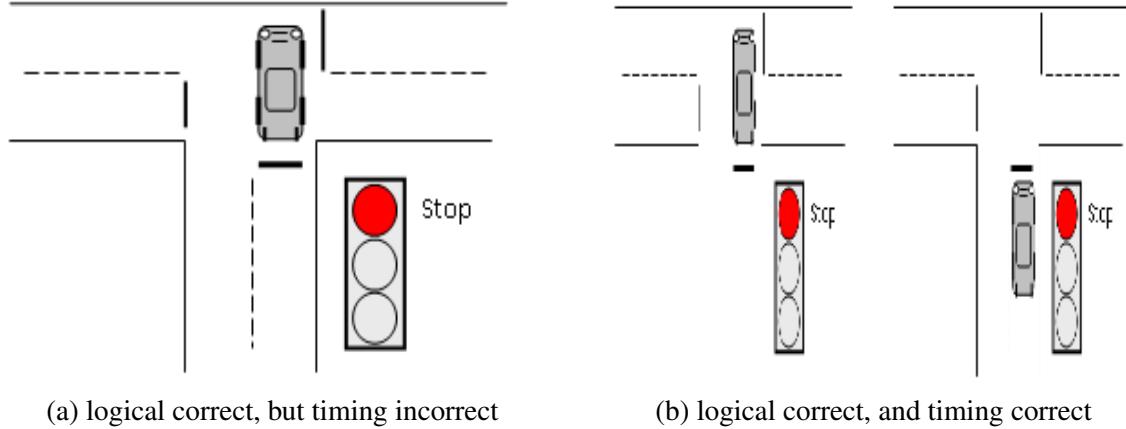


Figure 1.1.2: Moore's Law

Software for Real-Time Systems differs substantially from software for generalpurpose computers (PC) due to the **requirements on the real-time behavior** !

Many methods of **classical software development** of non-real-time systems can not be used due to the lack of predictability.

For the development of real-time systems, special methods are required.



1.2 Requirements

The validity of an operation of a real-time system depends on

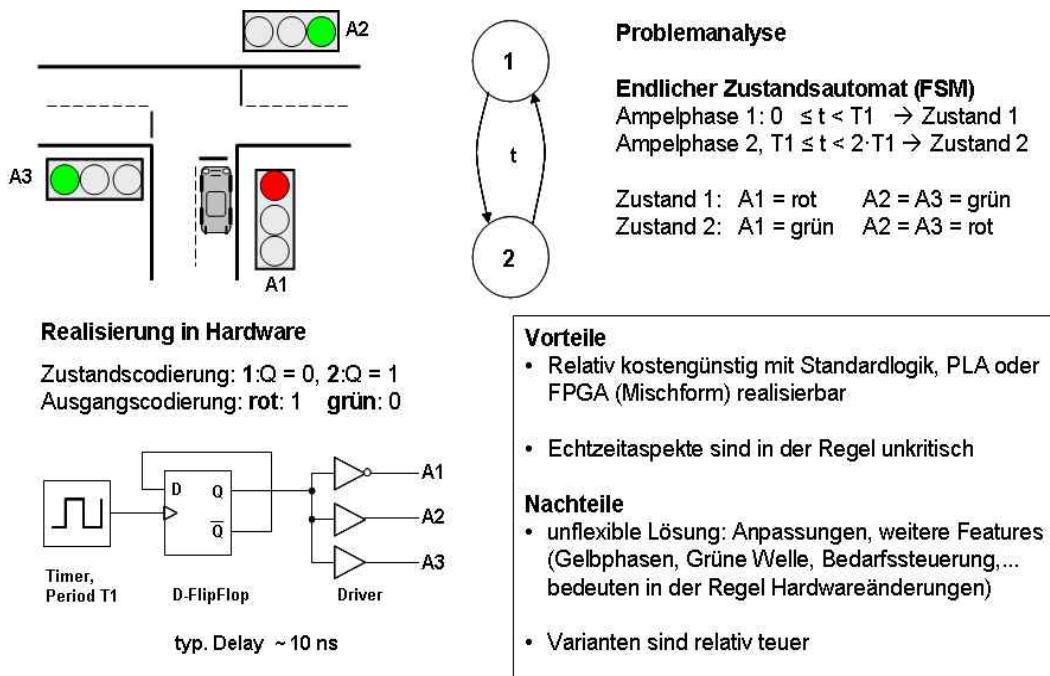
1. its logical result, and
2. the physical time this result is available

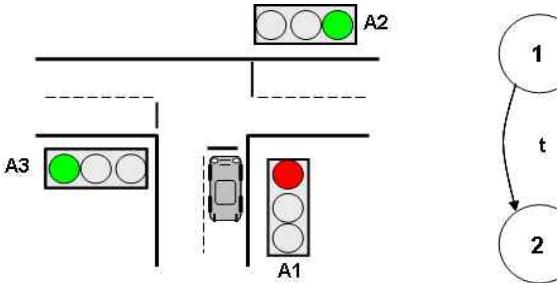
The traffic light control must determine traffic light phases logically and temporally correct.

For real-time systems, the logical correctness **and** the timing correctness is required.

- **Non Real-Time Systems:** logical correctness **OK**
- **Real-Time Systems:** logical correctness + timing correctness **OK**

Example: Real-Time Requirements with a Traffic Light Control





Problemanalyse

Endlicher Zustandsautomat (FSM)

Ampelphase 1: $0 \leq t < T_1 \rightarrow$ Zustand 1
Ampelphase 2, $T_1 \leq t < 2 \cdot T_1 \rightarrow$ Zustand 2

Zustand 1: A1 = rot A2 = A3 = grün
Zustand 2: A1 = grün A2 = A3 = rot

Realisierung in Software (FSM)

Zustandscodierung: 1:state = 0, 2:state = 1
Ausgangscodierung: rot: 1 grün: 0

```
// called periodic with T1
void AmpelTask() {
    static int state = 0;
    state = !state; // state change
    if (state == 0) {
        A1 = 1;
        A2 = A3 = 0;
    } else {
        A1 = 0;
        A2 = A3 = 1;
    }
}
```

Vorteile

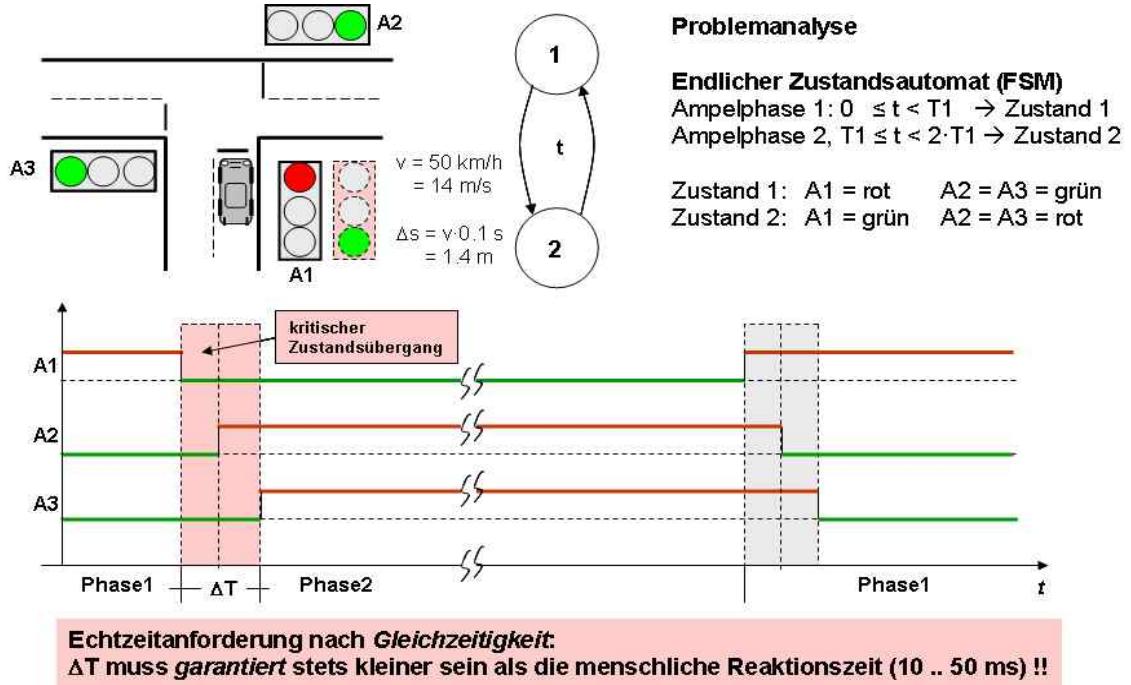
- Sehr kostengünstig mit µC realisierbar
(Embedded System mit/ohne Vernetzung „IoT“)
- Standard-Hardware + Embedded Software (C/C++)
- Flexible Lösung: Anpassungen, weitere Features
(Gelbphasen, Grüne Welle, Bedarfssteuerung,...
ohne Hardwareänderungen per Software-Update)

Nachteile

- Echtzeitanforderungen müssen berücksichtigt werden → **Echtzeitprogrammierung**

Solution: Critical Section (Mutex) in AmpelTask: see FSM Exercise

```
1 // called periodic with T1
2 void AmpelTask() {
3     static int state = 0;
4     state = !state; // state change
5     begin(mutex);
6     if (state == 0) {
7         A1 = 1;
8         A2 = A3 = 0;
9     } else {
10        A1 = 0;
11        A2 = A3 = 1;
12    }
13    end(mutex);
```



A Definition of Real-Time Systems is given by DIN 44300 [1985]

Realzeit-Systeme beziehungsweise Echtzeitsysteme sind Computersysteme, die im Realzeitbetrieb arbeiten.

Realzeitbetrieb wird definiert als der Betrieb eines Rechensystems, bei dem Programme zur Verarbeitung anfallender Daten ständig betriebsbereits sind, derart, dass die Verarbeitungsergebnisse innerhalb einer vorgegebenen Zeitspanne verfügbar sind. Die Daten können je nach Anwendungsfall nacheiner zeitlich zufälligen Verteilung oder zu vorherbestimmten Zeitpunkten anfallen.

Wikipedia:

In computer science, **real-time computing** (RTC), or reactive computing, is the study of hardware and software systems that are subject to a "real-time constraint", i.e., operational **deadlines** from **event to system response**.

By contrast, a **non-real-time system** is one for which there is **no deadline**, even if fast response or high performance is desired or preferred.

The needs of real-time software are often addressed in the context of real-time operating systems, and synchronous programming languages, which provide frameworks on which to build real-time application software.

A real time system may be one where its application can be considered (within context) to be mission critical.

The anti-lock brakes on a car are a simple example of a real-time computing system - the real-time constraint in this system is the short time in which the brakes must be released to prevent the wheel from locking. **Real-time computations** can be said to have **failed** if they are **not completed before their deadline**, where their deadline is relative to an event.

A real-time deadline must be met, regardless of system load.

Thus, in a real-time system the logical correctness (functional correctness) of the answers with guaranteed response times (temporal correctness) is of essential importance.

The temporal behavior in real-time systems is part of the system specification and thus subject of the product verification.

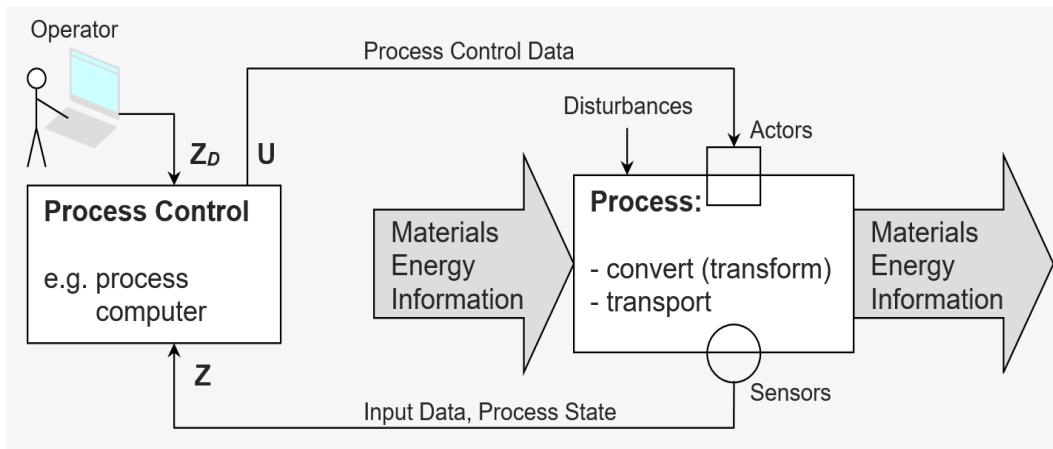
Embedded systems often have to meet **real-time requirements**.

Technical Process and Process Model

A technical process can be defined as follows (DIN 66 201):

In a technical process materials, energy or information (the process media) are converted or transported (such as an end product) into grafted materials, energy or information.

Its input and state variables summarized in the vector \mathbf{Z} can be measured with sensors and controlled by actuators:



In automation, the process is controlled via control actions (vector U) so that after a given time, a desired process target or a desired state Z_D is reached.

The desired state can refer to the following:

quality, quantity (of the final product), speed, stability, security (during the transforming process), production, yield, raw material consumption, emissions, or even reaching the next process step (e.g. with sequential processes).

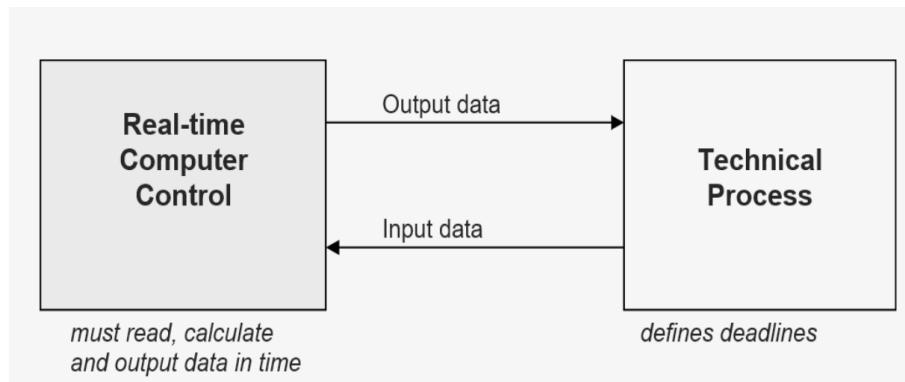
The delay time in the loop is of **crucial importance** for the **stability** of the **entire system**!

This delay time must therefore be exactly specifiable **Real-Time Programming** !

Timeliness

The timeliness (=timing correctness) is directly derivable from the above considerations requirement for all real-time systems. For digital process computers timing correctness is, the output data must be calculated in time.

This also requires that the input data must be collected on time. The timing deadlines, and are usually given by a technical process:



Thus the allowed response time for the controller is limited to a known value, thus real-time systems are said to be **predictable or deterministic**.

Input data must be read in time for a real-time system, and the output data to be generated must be provided within the given deadline (time condition).

The primary requirement for a real-time control is the **constant ability to input, process and output data on time regardless of the system workload**.

For a real-time system also means that it **monitors the temporal validity** of information and prevents the use of invalid data !

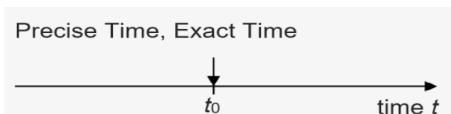
This must apply regardless of whether the data to be processed are **event based** or at **predetermined (e.g. periodic) times**.

However, due to complexity, the timing behavior of real-time systems **can never be predicted with 100 % certainty** System validation by tests !

Time Conditions

The **time conditions** to monitor, and control a process can be in several variants

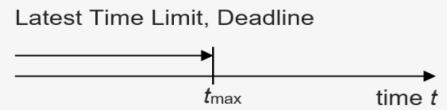
1. A precise time t_0 for a controller action is defined. This action must not be executed earlier or later.



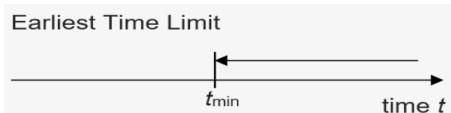
Examples: Sampling Systems (DSP & Digital Control), clock display control.

2. A maximum time t_{max} (=deadline) for a controller action is defined. This action must be finished latest at its deadline, but it can be finished earlier.

Examples: a sampling system calculating an output sample for the next sampling period, Anti-Squeeze Control, maximum response time with switching on/off a machine, UI control reaction time to a button, slider, ... e.g. $t_{max} < 50\text{ms}$ with human reaction times "immediately"



3. Earliest Time Limit A minimum time t_{\min} for a controller action is defined. The action must not be executed earlier, but it can be executed later.



Examples: transitions in state machines, an output may not occur before a state transition has been finished.

4. Time Interval An action must be executed within a time interval $[t_{\min}, t_{\max}]$, thus at any time earlier than t_{\max} and later than t_{\min} .



Example: airbag ignition in some crash situations an airbag shall be ignited e.g. 10-30 ms, in other situations 5-8 ms after crash.

Time conditions can be **periodic** or **non-periodic**.

Periodic Time Condition

A periodic time condition is given with analog signals to be sampled and processed by a digital computer. Any analog signal has to be discretized in both value and time by a constant periodic sampling time T_s . This is typical for sampling systems realizing digital filters or digital control algorithms for processing quasi analog signals.

An example is the electric signal of a microphone which has to be sampled **strictly periodic** in order to have a **digital representation** of the **original analog sound signal**. Any deviation from the periodic sampling results in unwanted noise, which cannot be compensated. After time sampling the values of the voltage signal is discretized by an AD-Converter.

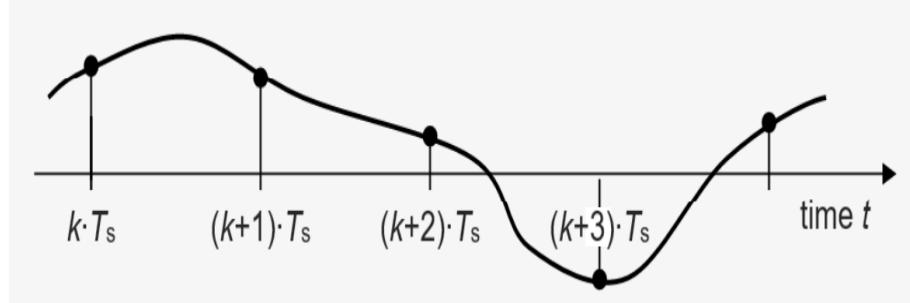
Sampling Theorem [Kamm], [Oppen]

The basic theorem of Shannon / Kotelnikov in signal theory states, that any analog signal $x(t)$, which is sampled at constant frequency f_s can be perfectly reconstructed from the sampling series $x(k \cdot T_s)$, apart from a constant delay, if the signal bandwidth B_x is not larger than half the sampling frequency $f_s = 1 / T_s$, so $B_x < f_s / 2$.

Example: CD-Audio: $f_s = 44.1 \text{ kHz}$, $T_s = 22.7 \mu\text{s}$ $B_x = 20 \text{ kHz} < 0.5 f_s = 22.05 \text{ kHz}$

Correct sampling: tagesschau_44.1.wav

Sampling theorem violation: tagesschau_D10.wav



Digital signal processing of analog signals involves very strict requirements to real-time conditions, due to the periodic and exact sampling conditions.

Action	Time Condition
input of a sample of an analog signal	exact periodic sampling at $k \bullet T_s$
computation of a signal sample	time interval $[k \bullet T_s + T_{ADC}, (k+1) \bullet T_s]$
output of the calculated signal sample	exact periodic sampling at $(k+1) \bullet T_s$

Non-Periodic Time Condition

A non-periodic time condition is given frequently with digital signals to be processed at arbitrary, non-predictable times due to events like pushing or releasing a switch or a revolution sensor, producing a slope at discrete angles of a motor shaft.



Absolute Time Condition

With an absolute time condition defined, an action must be executed at a certain global time, i.e. an action must be performed at 12:00:00 MEZ.

Relative Time Condition

With a relative time condition defined, an action must be executed relative to some previous event. Example: a control signal must be updated after 0.5 s after a push button was pressed.

All time conditions can arise in combinations. For meeting the time conditions a real-time system must have

1. **sufficient processing speed** to process the input and output data at the required speed (sampling frequency), which the technical process requires
2. **deterministic time behavior**

1.2.1 Hard and Soft Real-Time Systems

In the previous section, we saw that computation must complete before reaching a given deadline. In other words, real-time systems have timing constraints and are deadline-driven. Real-time systems can be classified, therefore, as either hard real-time systems or soft real-time systems.

What differentiates hard real-time systems and soft real-time systems are the degree of tolerance of missed deadlines, usefulness of computed results after missed deadlines, and severity of the penalty incurred for failing to meet deadlines.

For **hard real-time systems**, the level of tolerance for a missed deadline is extremely small or zero tolerance. The computed **results after the missed deadline** are likely **useless** for many of these systems. The penalty incurred for a missed deadline is **catastrophic** (e.g. an unstable control loop).

For **soft real-time systems**, however, the level of tolerance is non-zero. The computed **results after the missed deadline** have a **rate of depreciation**. The **usefulness** of the results does **not reach zero immediately** passing the deadline, as in the case of many hard real-time systems. The physical impact of a missed deadline is **non-catastrophic**.

1. Hard Real-Time Conditions

The time conditions must be met, or catastrophes occur ! Deviations from the time-conditions are not allowed and represent a serious error. In safety-related systems a violation of the hard real-time time conditions can endanger human life (e.g. if sampling controls become unstable).

Systems, meeting the hard real-time conditions are called ***Hard Real-Time Systems***.

Examples: Sampling Controls (closed-loop), aerospace flight controls, airbag ignition, automotive X-by-wire, pacemakers, ...

2. Soft Real-Time Conditions

The deadlines must be met but with a degree of flexibility. The deadlines can contain varying levels of tolerance, average timing deadlines, and even statistical distribution of response times with different degrees of acceptability. A missed deadline does not result in system failure, but costs can rise in proportion to the delay, depending on the application. Systems meeting soft real-time conditions are called ***Soft Real-Time Systems***.

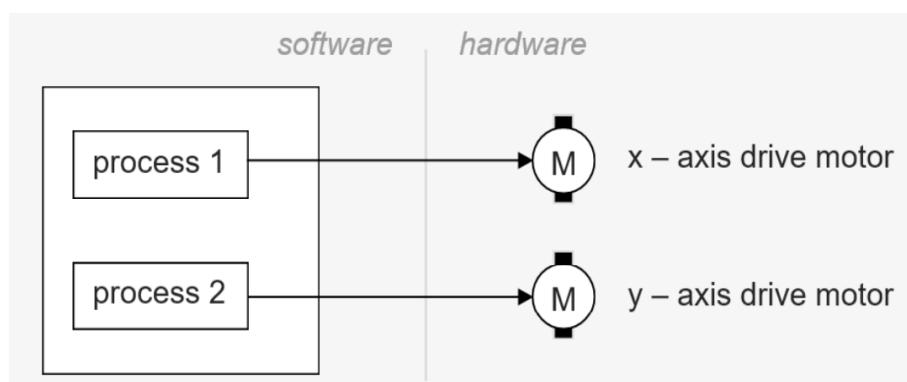
Examples: multimedia-streams (speech, video, music, low level audio), toasters, refrigerators,

...

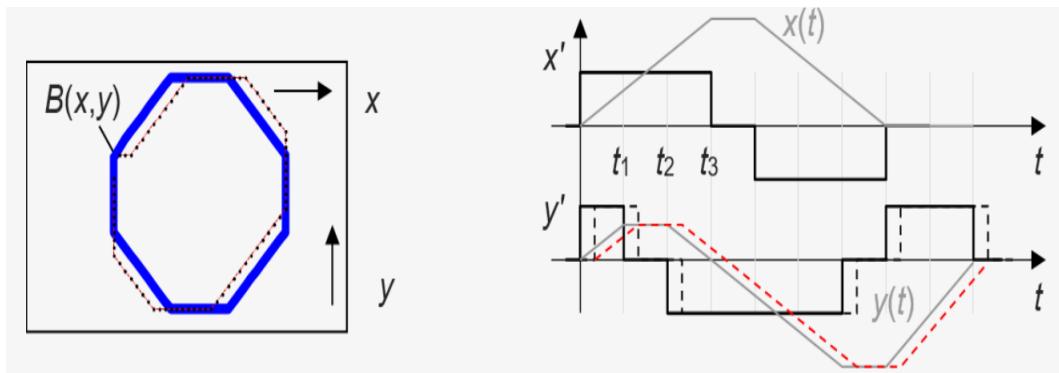
1.2.2 Concurrency

Real-time systems must generally treat multiple inputs and outputs simultaneously. Thus, the simultaneous addition to the timeliness of the second general requirement for real-time systems.

Example is about a CNC machine tool, where the x-y axes must be controlled simultaneously (synchronously, concurrently) to move the tool along a predetermined path $B(x, y) = B(x(t), y(t))$.



There is a relative time condition for both processes acting concurrently on the two drive axes x and y:



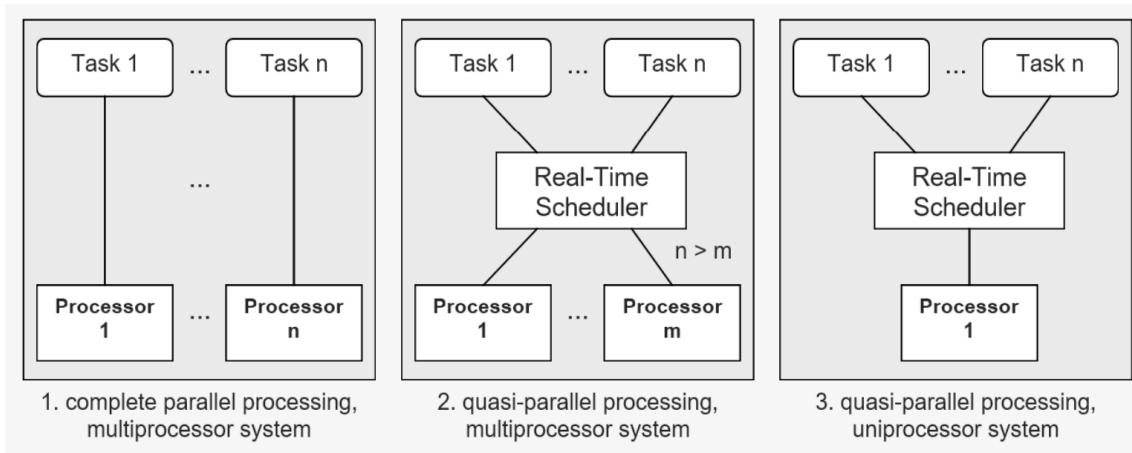
If the time axis of x and y are shifted by some amount of time ΔT (e.g. due to different clocks for x- and y-drive control), the resulting path $B(x, y) = B(x(t), y(t-\Delta T))$ will be wrong.

Example: Tolerances $|x|, |y| < 0.05$ mm, Requirement $v_x = v_y = x' = y' = \pm 10$ cm/s Requirement $|t| < |x| / |v_x| = 0.5$ ms Synchronization, Real-Time Condition

1.2.3 Realizations of Real-Time Systems with Concurrency Requirement

To meet the requirement of concurrency, there are several possibilities:

1. Full parallel processing in a multiprocessor system
2. Quasi-parallel processing in a multiprocessor system
3. Quasi-parallel processing in a uniprocessor system (**Multitasking**)



Each task is processed on a separate processor (e.g. microcontroller). There is a real parallel processing, each task has the full power of its own processor. This allows the independent consideration of each task, however, the tasks must be synchronized in most cases (e.g. x-y axis control of the CNC machine).

Each task executed on a processor that is allocated by a real-time scheduler (there are usually fewer processors available than tasks). The real-time scheduler is a hardware or software component, which assigns the tasks to processors such that all tasks can meet their timing constraints. The individual tasks can be assigned to single processors.

From a hardware perspective, the simplest and most cost effective realization: All tasks share a single processor **Multitasking**. A Real Time Scheduler controls the allocation of the processor to the task. This form of real-time scheduling is the easiest to analyze and control

→ preferred realization of most embedded systems.

Availability:

Real-time systems must be available without interruption, otherwise time conditions may be violated. This leads to the third basic requirement of real-time systems, the **availability**.

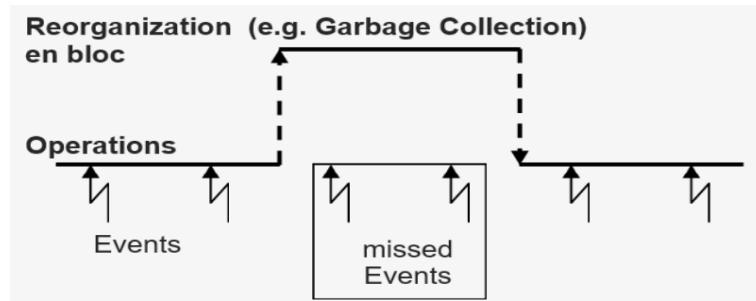
Real-time systems must be available over a longer period of time, maybe around the clock without any breaks, as with

- Power Plant Controls

- Production Facilities
- Heating and Air Conditioning Systems
- Communication
- Medical Applications ...

This requires that there is no disruption of operations for phases of system reorganization. A typical example here is the garbage collection of some programming languages such as Java or C # or their runtime environments.

For example, the default Java implementation is not suitable for use in real-time systems.



Remedy is the use of algorithms free of need for reorganization (such as for dynamic memory management).

Example : Algorithm for Calculation of a Factorial in C

```
/* reorganization free algorithm
   for calculation of a factorial */

int fact(int n)
{
    int f = 1;
    for (i = 1; i < n; i++) {
        f *= i;
    }
    return f;
}
```

```
/* algorithm needs reorganization
   with calculation of a factorial */

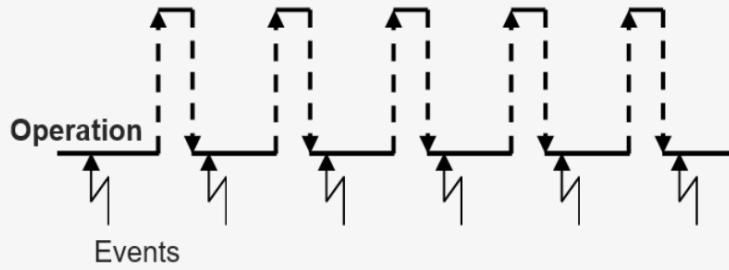
int fact(int n)
{
    if (n > 1)
        return n*fact(n-1); // dynamic
    else
        return 1;           // memory
                           // needed here
}
```

OK with Real-Time programming

Not OK with Real-Time programming ! (Exec.-Time of stack operations not deterministic due to recursion).

Another possibility is to divide the reorganization into small steps under control of the real-time scheduler to ensure that no time constraint is violated:

**Reorganization in small steps,
controlled by the real-time scheduler**



This method has been used by Real-Time Java (Real-Time Specification for Java (RTSJ)). Also, more traditional memory management strategies involve non predictable time for execution (like **malloc/free** of the C standard library, **new/delete** with C++)

→ dynamic memory management is typically avoided with hard real-time systems !

Example:

```
/* dynamic memory management:  
not real-time capable due to  
stdlib: malloc/free */  
  
int foo(int n) // periodic call  
{  
    int *buf = (int*)malloc(n);  
    int x = -1;  
    if (buf) {  
        x = some_algorithm(buf, n);  
        free(buf);  
    }  
    return x;  
}
```

```
/* static memory management:  
fully real-time capable */  
  
int buf[NMAX]; // static alloc  
  
int foo(int n) // periodic call  
{  
    int x = -1;  
    if (n <= NMAX) {  
        x = some_algorithm(buf, n);  
    }  
    return x;  
}
```

Not OK with Real-Time programming
(Exec.-Time of malloc() non deterministic)
- but: malloc() in initialization possible

OK with Real-Time programming !

1.3 Timing Definitions

Process Time and Processing Time

The time interval between two requests of the same type is called the **process time** T_P . If the interval between two events is constant, it is called a periodic event. In many cases, however, the time interval varies, so that there is a time interval with a minimum T_{Pmin} and a maximum T_{Pmax} .

For the actual processing of the event computer instructions (operations) have to be executed. The event belongs to the sequence of instructions is referred to as the code sequence or a job. Jobs are implemented within the computer as a computational process, which can be distinguished as **tasks** and **threads** (see section **Process Management**).

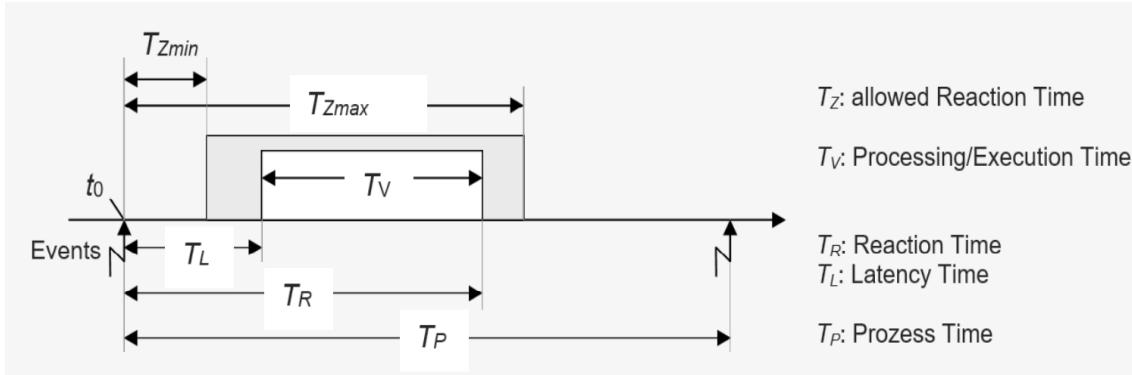
For the execution of a job i the computer needs **processing time** T_{Vi} (from de: *Verarbeitungszeit*) which is also called **execution time**.

The **processing/execution time** needed for a certain number of operations N is inverse to the **computing power P** (in ops / sec) a processor core (CPU) offers:

$$T_v = N/P \quad [T_v] = \text{ops}/(\text{ops/sec}) = \text{sec} \quad (1.1)$$

The CPU power is approximately proportional to the CPU clock frequency.

In practice it is not easy to specify the processing/execution time for a certain task, since it often depends on the input values (conditional branches). Furthermore, the performance of the processor varies (due to caches, pipelines and DMA). To calculate T_{Vmax} in real-time systems, N_{max} and P_{min} shall be used in 1.1.



Timeliness, Response time and Maximum Response Time

Timeliness means, that a task required at the time t_0 is done **not before** a specified time $t_0 + T_{Zmin}$ and is done **no later** than a specified time $t_0 + T_{Zmax}$ (timeliness).

The requirement for a minimum latency, i.e., that a task is not been done before a certain time, is often missing ($T_{Zmin} = 0$) or can be realized easily.

On the other hand, the requirement for **timeliness** ($T_R < T_{Zmax}$) is hard to guarantee.

The **maximum reaction time** T_{Rmax} is the worst-case time when task execution ends and the computer can cause a reaction.

The **maximum reaction time** T_{Rmax} always needs to be **smaller** than the **maximum response time** T_{Zmax} .

Utilization, Load

The computer core (CPU, processor) is loaded (utilized) depending on the frequency of occurrence of an event. The utilization gives a relative measure how much the CPU is loaded by a task is the quotient of the necessary processing time T_V and process time T_P

$$\rho = T_v/T_p \quad (1.2)$$

The total utilization ρ_{ges} is the sum of the utilization of all individual jobs 2.2. A real-time system must be able to process all tasks with fulfilling their requirements for timeliness – this is sometimes referred to as the requirement for concurrency (in a wider sense).

Mathematically, this means that the total utilization ρ_{ges} is less than 100%:

$$\begin{aligned}\rho_{ges} &= \sum_{i=1}^n \rho_i = \sum_{i=1}^n \frac{T_{Vi}}{T_{Pi}} \\ \rho_{ges} &< 1\end{aligned}\tag{1.3}$$

1.4 Real-Time Evidence

Evidence that all time requirements of all requests are met under all conditions, is very difficult (practically impossible).

In principle, the real-time evidence is based on

1. evaluate the relevant characteristics of the technical process,
2. number of different requirements
3. Minimum processing time for each request
4. Minimum allowable response time for each request T_{Zmin}
5. Maximum allowable response time for each request T_{Zmax}
6. dependencies between events
7. identify the maximum processing time T_{VMax} (WCET = Worst Case Execution Time) for each request
8. check the **load condition** (2.2)
9. verify the **timeliness condition** (determination of T_{Rmin} and T_{Rmax}).

With verifying the **timeliness condition** in particular, it is not trivial to determine the maximum response time T_{Rmax} .

Estimate for the Worst Case Execution Time (WCET)

The maximum processing time T_{VMax} – often called Worst Case Execution Time (WCET) – of a request is very difficult to determine. The WCET depends in particular on the algorithm, the implementation of the algorithm, the hardware used and the other activities of the system (other computing processes).

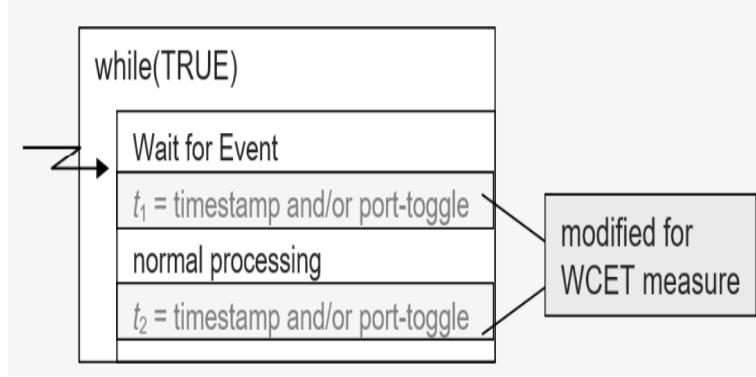
Therefore, the WCET at best be estimated. In principle, two methods for determining the WCET can be distinguished:

1. measure the WCET
2. static code analysis with WCET determination (count operations,cycle).

WCET Measure

The WCET is determined by execution of the code sequence that will be processed, normally on the target platform. The time between the onset of the event and output at the end of the code sequence is determined.

Condition: Modification of the real-time software



Method for Measuring WCET

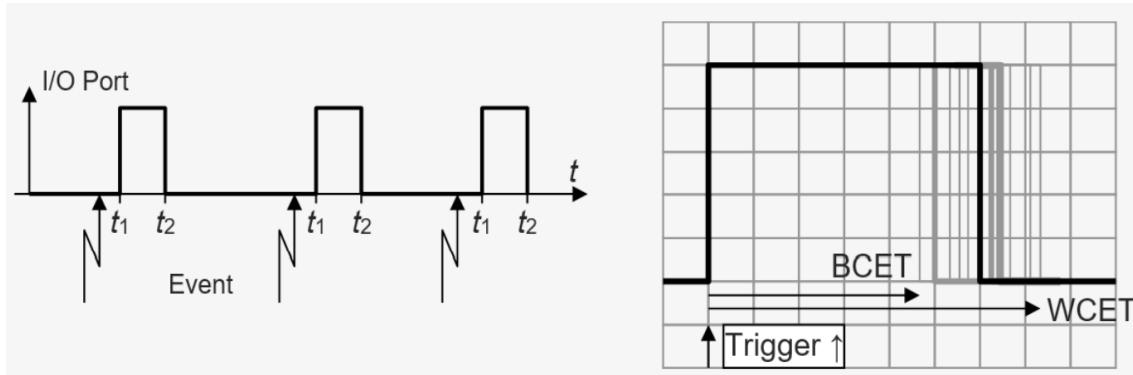
To measure this time can be divided into two basic methods:

1st Measurement by the measure out piece of code itself (see above). The piece of code will be amended such that every time an event arrives and when the reaction took place, a time stamp t_1 (free running timer) is stored. At the end of the code, a second time stamp t_2 is stored.
 → the difference $T_{Vi} = t_2 - t_1$ is the actual execution time.

The first value T_{V1} is stored as T_{Vmax} . With subsequent iterations, a newly calculated time T_{Vi} is compared with the stored value. If the newly calculated time is greater than the previously measured WCET, the new value overwrites the existing WCET

$$T_{Vmax} = \max(T_{Vi})$$

External measurement of event and response (output), by means of an oscilloscope using port-toggle.



For a WCET measure, appropriate input values and events must be generated. In addition, the entire system must be put under load (e.g. by an automatic tester).

Advantages of the method:

1. Independent of a programming language
2. Relatively easy to implement
3. Can run as autonomous self-diagnosis

Disadvantages of the method:

- The WCET of a code sequence can not be guaranteed, after all, it depends on many factors (background, loop iterations, caches, branches, etc ..).
- The measurement is WCET is time consuming and ultimately expensive. Measure out the piece of code needs to be theoretically charged with all input data (in any combination).
- This method can be carried out with the running code on the target platform only. Thus, an assessment at an early stage of development can be difficult.
- A test environment is required (which provides the input data to create).
- The code sequence to be measured out must be modified (for storing time stamps) → **timers**.
- It is sometimes difficult to put the system under the required load.

1.4.1 Static Code Analysis

Here, the code itself is analyzed. Therefore most often an analysis tool program is used, which is fed with the object (machine) code, rather than with the C-source code. In addition, a hardware description is necessary (e.g. clock frequency of the micro-controller).

The tool allows the duration of each code sequences to be estimated. After that, the number of loop iterations are considered, a bound of the number of cycles and the estimated execution time T_{SA} is printed.

Analysis can get rather complex, if cache hits, cache misses, processor pipelining are considered.

If there are safety requirements, the code measured is not to changed after a static code analysis.

As a safety margin factors of $c = 2, 3$, sometimes are used to state an upper limit

$$t_{Vmax} \leq T_{SA} \cdot c$$

A manual determination of the number of cycles can be done in simple cases by inspection of the C-compiler generated assembler code:

Example: timer-interrupt routine for AVR microcontroller (e.g. ATmega8_sum.pdf).

```
// ISR() is called with frequency
// FS * TSCNT
ISR(SIG_OVERFLOW0)
{
    // LED 1 blinks at PORTB
    if (cnt < (TSCNT>>1)) // TSCNT / 2
        PORTB |= 1;           // LED 1 on
    else
        PORTB &= ~0x01;      // LED 1 off

    cnt++;
    if (cnt == TSCNT) cnt = 0;
}
```

```
if (cnt < (TSCNT>>1)) //
a4: 20 34          cpi   r18, 0x40 ; 64
a6: 10 f4          brcc  .+4      ; <_stack+0xd>
    portb |= 1;    // LED 1 on
a8: 91 60          ori    r25, 0x01 ; 1
aa: 01 c0          rjmp  .+2      ; <_stack+0xf>
    else
        portb &= ~0x01; // LED 1 off
ac: 9e 7f          andi   r25, 0xFE ; 254
    cnt++; if (cnt == TSCNT) cnt = 0;
c2: 82 2f          mov    r24, r18
c4: 8f 5f          subi   r24, 0xFF ; 255
c6: 80 93 60 00    sts    0x0060, r24
ca: 80 38          cpi    r24, 0x80 ; 128
```

The C-compiler generated assembly-/machine-code, ultimately determines the number of cycles to complete a section of code for a particular microcontroller (AVR here). The number of cycles multiplied by the instruction cycle time is the execution time.

Estimation of the Best Case Execution Time (BCET)

The determination Best Case Execution Time of a job can be done as the WCET, with the least possible load on the system, which can be easily implemented in general-at least in a test environment.

$$T_{Vmin} = \min(T_{Vi})$$

Chapter 2

Real-Time Operating Systems

An **RTOS** (Real-time Operating System) must meet the same requirements as a standard general purpose operating system (GPOS) and offers services for

- **Task Management:** This is the management and organization of the implemented programs to be processed, also called tasks. The mission of the task management is thus essentially in the allocation of the processor (or the processors in a multiprocessor system) to the task.
- **Resource Management:** Tasks need resources for their execution, their allocation is the task of resource management. This mainly includes:- Memory management, responsible for allocating memory- Input/Output (I/O) management responsible for the allocation of I/O devices to the tasks.
- **Communication:** The communication between tasks, called inter-process communication.
- **Synchronization:** A special form of communication is the synchronization, which refers to the timing of the tasks.
- **Protection:** The protection of resources against unauthorized access by tasks.

These traditional requirements are exactly the same RTOSes as with GPOSes. Depending on the application, some of the services might be implemented only rudimentary or completely absent. This is in particular with **embedded systems**, where the **RTOS** is kept as **lean as possible**, due to the limited resources.

In addition to the traditional OS requirements, real-time operating systems are required to

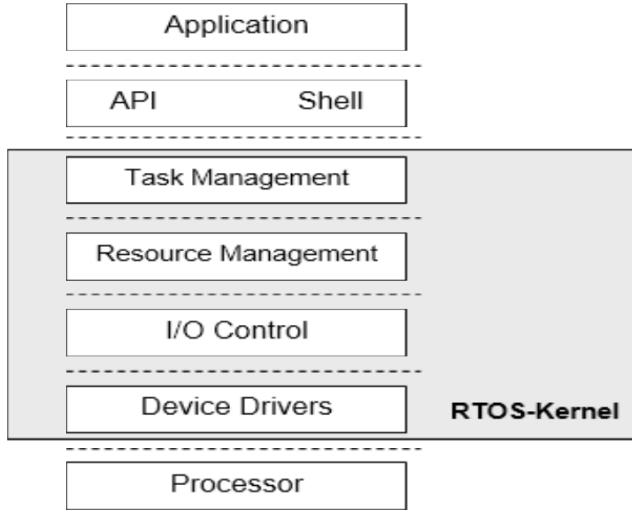
- Respect for **timeliness** and **concurrency**
- Respect for **availability**

These requirements of RTOS dominate the other requirements, even if compromises are necessary.

2.0.1 Structure of an RTOS

A real-time operating system (RTOS) is a program that schedules execution in a timely manner, manages system resources, and provides a consistent foundation for developing application code.

Early operating systems had a monolithic structure, i.e. all functionality was implemented in a uniform, not further subdivided software block. This led to a number of disadvantages such as poor



maintainability, poor adaptability and high error rate.

Today's operating systems therefore follow a hierarchical layers model.

- **Device Driver:** This layer abstracts from the hardware, and
 - realizes the **hardware-dependent control** of each device
 - realizes the **hardware-independent** interface for the above layer.

Ideally, the device drivers are the only hardware dependent layer in the OS. When adapting to other devices, only the device driver layer must be changed.

- **I/O (Input-/Output) Control**, realizes the **logical, hardware-independent** device control.
- **Resource Management:** responsible for (allocation) and de-allocation (release) of memory and I/O resources.
- **Task Management:** Responsible for the allocation of the processor to each task.
- **API (Application Program Interface):** Realizes the interface to the application.

The OS kernel is critical for the stability and security, it is executed in the so-called **kernel mode** of the processor, which allows full access to all resources.

The **kernel mode** is a special operating mode of (advanced) microprocessors, enabling privileged instructions, direct access to memory and I/O, change configuration registers, etc.

In normal **user mode** these privileged commands are blocked so that an application can not interfere with important operating system parts. This is one of the protection services of the operating system.

In the previous layer model, the core operating system extends over the layers 1 – 4. Since the core contains many layers, it is called a **macro core operating system**.

Today's RTOSes must be highly configurable, especially in the field of **embedded systems** where scarce resources are typical → FreeRTOS.

It is therefore desirable to remove unwanted parts from the operating system, e.g. not needed scheduling or protection methods. This leads to the concept of the **micro-kernel operating system**.

2.1 Task Management

The task management is a core task of operating systems. The most significant differences between standard and real-time operating systems are found here.

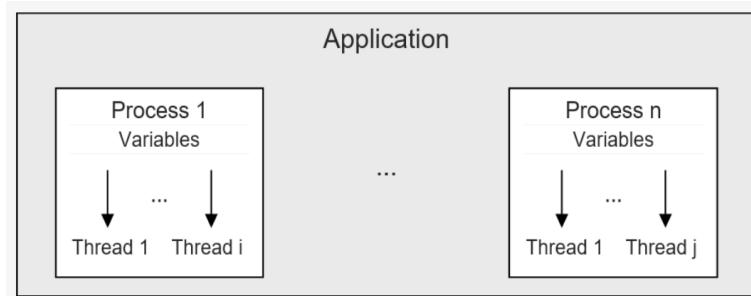
The tasks in a real-time application must meet the requirements for **timeliness** and **concurrency**. This requires scheduling strategies for RTOSes different from those found in standard operating systems.

Task Model

A **computational process** or **process** (also called **task**) is running as a computer program together with all the variables (including register states) and resources → each process has a **main()**

A task is a process controlled by the RTOS for execution of a sequential program. Several tasks are being processed by the quasi-parallel processor, necessary changes between tasks are made by the RTOS' task scheduler.

Changes between tasks are needed to meet all tasks requirements for timeliness.



Within a **process**, there can be parallel **threads** (running simultaneously).

A **thread** must share its resources with other **threads** of the same **process** → just 1 **main()**

A **task** is called a **heavyweight process**, if it contains its own variables and resources separated from other tasks by the OS. It has its own address space and can communicate with other tasks via inter-process communication.

→ A task realized as a process provides maximum protection, possible interference by other tasks is limited to predefined channels.

→ A change of the processor to another task (**context switch**) is due to the separate resources, which is a time-consuming task.

A **task** realized as a **thread** is called a **lightweight process** that exists within a single process. It uses the variables and resources of the process. All threads within a process **share the same address space**. Communication can take place over any global variable within the process.

- Threads can interfere with any other thread within a task.
- Shared memory allows great efficiency.
- Communication between threads is more direct and faster

- Context switch can take place very quickly, e.g.: FreeRTOS, VxWorks
- Low data protection between data of individual threads

To fulfill the **real-time requirements**, efficiency is usually more important than protection.

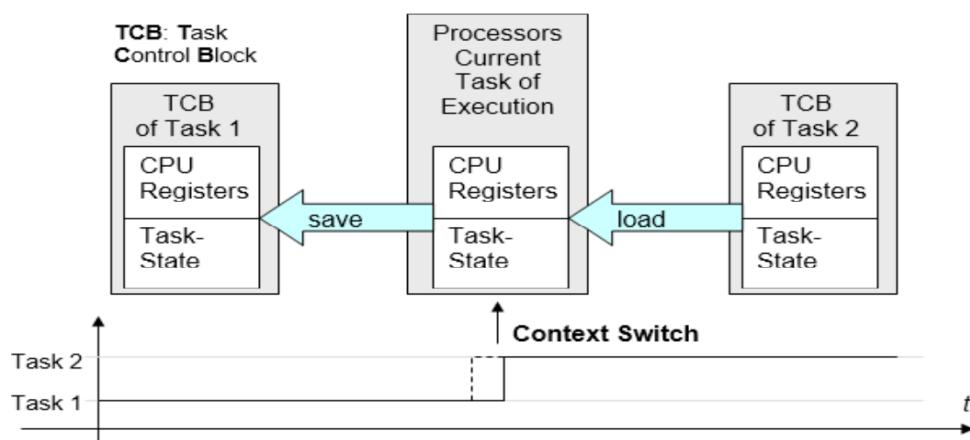
→ Many real-time applications use **threads within a single task**.

Often, a real-time application is realized by a **single process**, which contains threads to realize numerous tasks. **Embedded systems** are often based on the **thread concept**, due to scarce resources.

From the perspective of the real-time conditions (timeliness, concurrency, availability), **threads** aren't different from **processes**, both are usually identified by the term **task**.

Multitasking, Context Switch

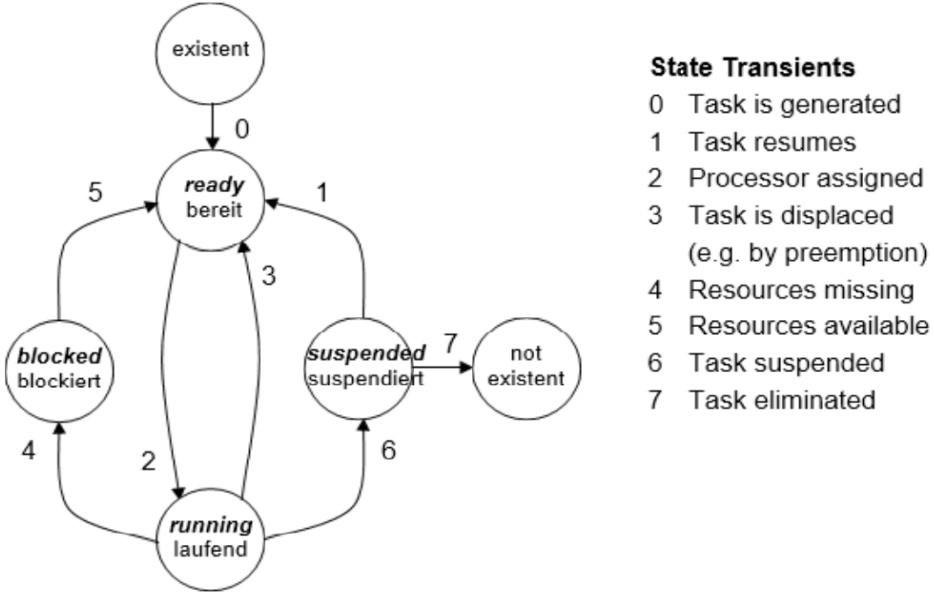
Multitasking is the ability of the OS to handle multiple tasks within set deadlines. An RTOS kernel might have 2 tasks that it has to schedule to run. At certain times, execution of Task 1 has to switch to Task 2



Task States

For each task (or threads) one can define 6 different states:

- **existent**
- **ready** (waiting)The task is ready, all conditions are fulfilled, resources are allocated, the task is waiting for the allocation of the processor.
- **running** (executing)The task is run on the processor. In a uniprocessor only one task can be in this condition, in a multi-processor system, several tasks can be in this state simultaneously.
- **blocked** (blocked)The task is waiting for an event (e.g. an input value, an inter-process communication object) or the release of a resource (task synchronization).
- **suspended** (finished)A task is suspended from its normal operations by another task. It can be resumed later
- **not existent**



2.1.1 Real-Time Scheduling

The main job of an RTOS task management is the allocation of the processor to the ready tasks. There are different strategies, so-called scheduling strategies.

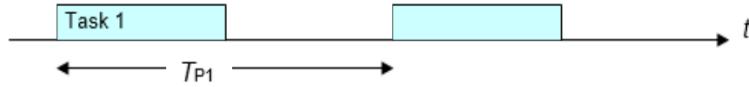
A **real-time scheduler** must divide all ready (runnable) tasks to the processor, so that all time conditions are met. The set of tasks managed by the real-time scheduler is called **taskset**.

For the evaluation of various scheduling strategies, the processor demand H (=utilization) is an important quantity for the load of the processor, it is defined as

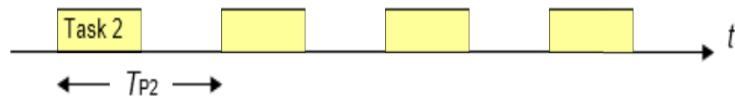
$$H = \frac{\text{DemandedProcessorTime}}{\text{AvailableProcessorTime}} \quad (2.1)$$

Each CPU can be utilized up to 100 % at maximum.

Example: A periodic task 1 with a period of $T_{P1} = 200$ ms and an execution time of $T_{e1} = 100$ ms causes a processor demand of $H_1 = 50$ %

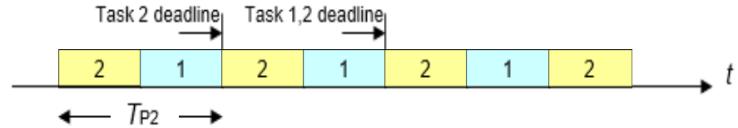


A second periodic task with a period of $T_{P2} = 100$ ms and an execution time of $T_{e2} = 50$ ms also causes a processor demand of $H_2 = 50$ %.

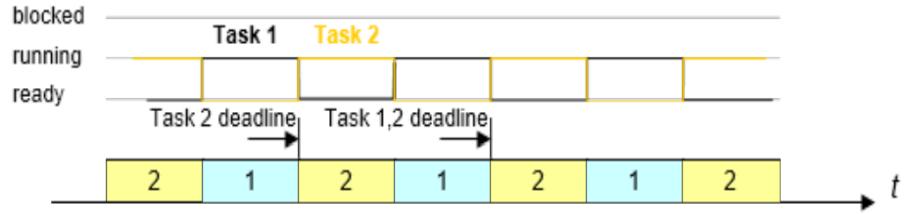


An implicit timing condition for each periodic task is: task execution must be finished before the next period starts (deadline) !

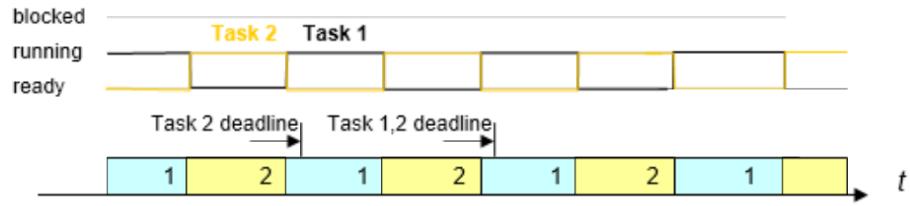
Both tasks cause a total processor demand $H = H_1 + H_2 = 100$ %.



One possible schedule for executing both tasks is to displace task 1 after exactly half of its execution time by task 2, which is never displaced. Thus, task 1 must be displaceable:



another possible schedule:

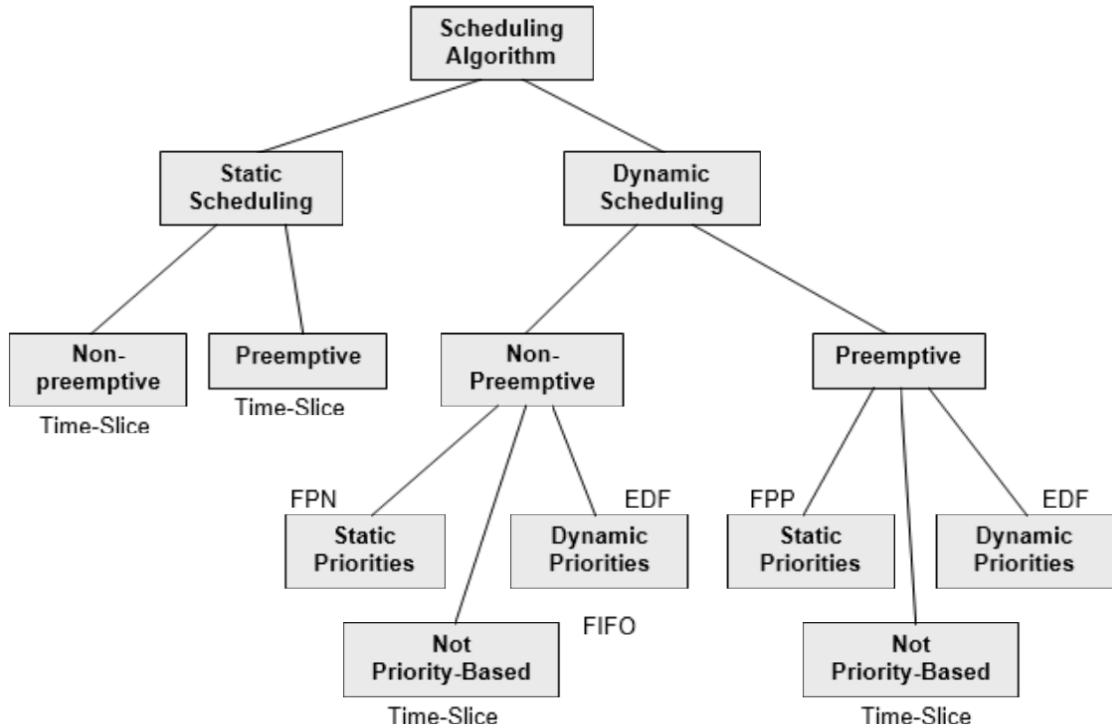


In general, for a tasklet of n periodic tasks, the total processor demand

$$H = \sum_{i=1}^n \frac{T_{ei}}{T_{pi}} \quad (2.2)$$

T_{ei} execution time of task i , T_{pi} periodic time of task i

Classification of Scheduling Algorithms



Static Scheduling

Prior to the execution of a taskset an allocation table (dispatching table) with the start times exists, regarding time constraints and dependencies of each task. The dispatcher as part of the RTOS kernel assigns the individual tasks due to this table
→ principle of synchronous programming.

Advantage: minimal overhead, as no decisions are needed at runtime.

Disadvantage: restriction to periodic events.

Dynamic Scheduling

Here for the execution of tasks different criteria are used by the dispatcher **at runtime**, taking into account start times, time conditions (deadlines), and dependencies of each task. Asynchronous programming. Advantage increased flexibility and the opportunity to respond to aperiodic events. Disadvantage increased overhead, lower predictability

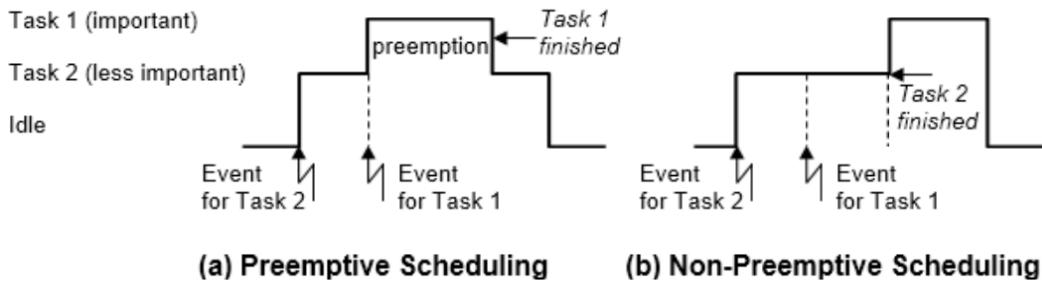
Static and Dynamic Priorities

(Not to be confused with static/dynamic scheduling). The scheduler can use priorities for the allocation of resources (processor, memory, I/O,...) to a task. **Static priorities** are set before running the application and are never changed during runtime. **Dynamic priorities** can be adjusted for each task **at runtime**. Furthermore, there are algorithms, using **No Priorities** at all.

2.1.2 Preemption

(= "Prioritization") is an important feature for a scheduler, which describes the capability for displacing a running task for later execution in favor of another task.

Preemptive scheduling means that a less important task can be displaced by a more important task. The most important task with **ready-state** will be executed immediately. The less important task will be continued again until no other more important task waits with **ready-state**.



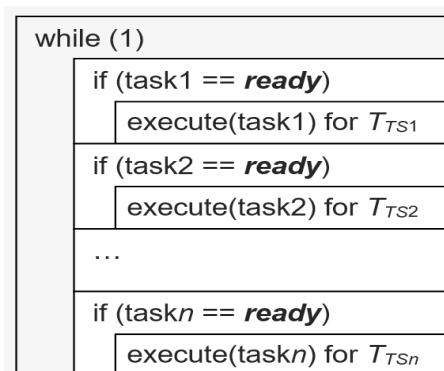
Non-preemptive scheduling (= **cooperative scheduling**) means, that there is no displacement of a running task. Only after the current task is finished or blocked, the next task with **ready** state is executed.

Time-Slice Scheduling (Round-Robin Scheduling)

Time-Slice Scheduling (time slicing) assigns each task a fixed time slice (Time Slice). The order of task execution corresponds to the sequence of **ready**-tasks entering the task-list of the OS scheduler (FIFO principle). The duration of the time slice for a task can be set individually.

This type of time-slice scheduling is a dynamic preemptive scheduling.

- TSS does not use priorities
- time slot duration can be chosen individually for each task
- With time slices chosen fine-grained enough TSS is approaching optimality.



Principle of Time-Slice (Round-Robin) Scheduling:

All **ready**-tasks are queued in a FIFO

Each task i gets assigned a time slice (time slice, quantum) T_{TSi}

- the task is preempted, i.e. displaced to the **ready** state
- the task is put at the end of the FIFO queue;
- The first task in the FIFO will be executed.

If a task changes state from **blocked** to **ready**, it will be put at the end of the FIFO queue

Typical Applications: Kernel-mode programs

Example 1: 3 Tasks with $H_{max} < 0.778$, from [wörnB], same as in section 2.2.7:

Task T1: period $T_{p1} = 10$ ms, execution time $T_{e1} = 1$ ms $\rightarrow H_1 = 0.1$

Task T2: period $T_{p2} = 10$ ms, execution time $T_{e2} = 5$ ms $\rightarrow H_2 = 0.5$

Task T3: non-periodic, period T_{p3} = deadline $T_{d3} = 15.4$ ms, $T_{e3} = 2.62$ ms $\rightarrow H_3 = 0.17$

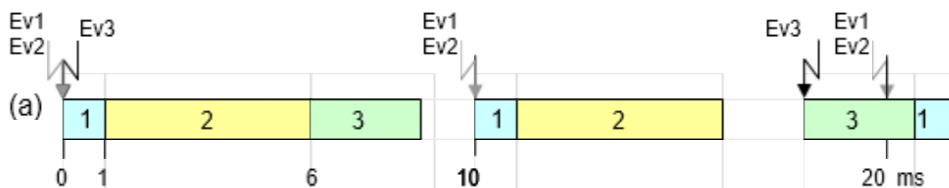
By (2.2) the total CPU utilization $H = 0.1 + 0.5 + 0.17 = 0.77$

One chooses a basic time-slice $T_{TS} = 1$ ms and assigns individual time slices as multiples of T_{TS} with $\sum T_{TS} = 9$ ms:

Task T1: $T_{TS1} = 1 \cdot T_{TS} = 1$ ms $\rightarrow H_1 = T_{TS1}/\sum T_{TS} = 1 \text{ ms} / 9 \text{ ms} = 11 \%$

Task T2: $T_{TS2} = 5 \cdot T_{TS} = 5$ ms $\rightarrow H_2 = T_{TS2}/\sum T_{TS} = 5 \text{ ms} / 9 \text{ ms} = 55 \%$

Task T3: $T_{TS3} = 3 \cdot T_{TS} = 3$ ms $\rightarrow H_3 = T_{TS3}/\sum T_{TS} = 3 \text{ ms} / 9 \text{ ms} = 33 \%$



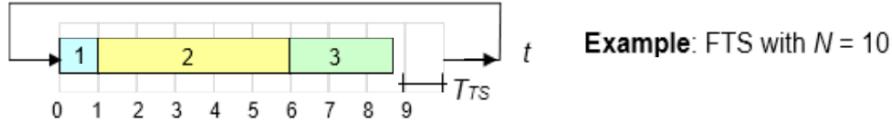
In this example the assigned time slices were chosen to be larger than the execution times of each task, in order to avoid preemption.

With TSS a context-switch can occur at the end of a time-slice only. which can cause some *jitter* (as with task T1 above).

Fixed Time-Slice-Scheduling

FT scheduling is based on a TDMA approach, requires no priorities, and is therefore frequently used in embedded microcontroller applications without an operating system.

A strictly periodic schedule is created with a periodic time $N \cdot T_{TS}$, i.e. with N integer multiples of the basic T_{TS} time slice. Each task gets assigned one (or more) basic time slots, large enough to hold the WCET of each:



Conditions for Events are polled within each task, if a task is not finished at the end of its timeslice, it can be preempted (*preemptive FTS*) or not (*cooperative FTS*).

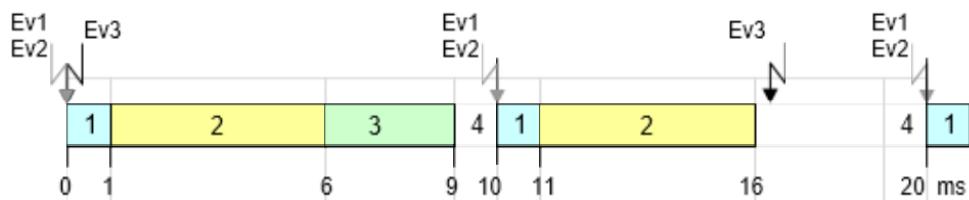
The previous example with three tasks, and an idle task T4 at a period $N \bullet T_{TS} = \Sigma T_{TS} = 10$ ms is realized, the time slices are chosen as multiples of $T_{TS} = 1$ ms as follows

Task T1: $T_{TS1} = 1 \text{ ms}$, $T_{TS} = 10 \text{ ms}$ $\rightarrow H_1 = T_{TS1}/\sum T_{TS} = 1 \text{ ms} / 10 \text{ ms} = 10 \%$

Task T2: $T_{TS2} = 5 \text{ ms}$, $T_{TS} = 5 \text{ ms}$ $\rightarrow H_2 = T_{TS2}/\sum T_{TS} = 5 \text{ ms} / 10 \text{ ms} = 50 \%$

Task T3: $T_{TS3} = 3 \cdot T_{TS} = 3 \text{ ms} \rightarrow H_3 = T_{TS3}/2\sum T_{TS} = 3 \text{ ms} / 10 \text{ ms} = 30 \%$

Task T4 (idle) $T_{TS4} = 1 . T_{TS} = 1 \text{ ms} \rightarrow H_4 = 10 \%$

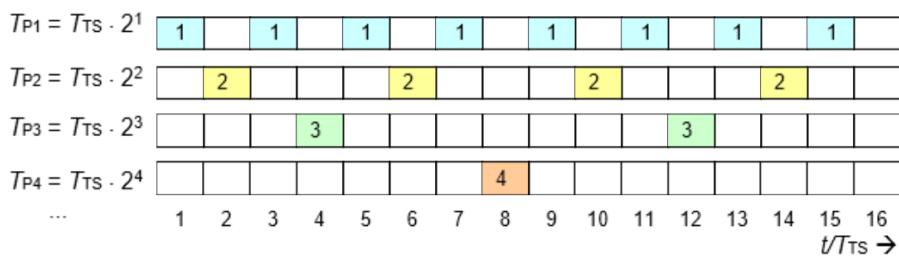


T3 runs for the second time at $t = 26$ ms the delay of 10 ms does not result in a T3-deadline violation. The idle task T4 provides a placeholder for unused CPU load, which can be used in modern CPUs for power saving features.

Advantage of fixed TSS: the execution times can be defined exactly!

Binary Fixed Time-Slice-Scheduling

If task periodic times T_{Pi} can be represented as power-of-2 integer multiples of a basic time slice T_{TS} , one can obtain a periodic binary schedule:



If the maximum time of execution is T_{TS} and equal for each task, the utilization for each task is

$$H_i = \frac{T_{ei}}{T_{pi}} = \frac{T_{TS}}{2^i \cdot T_{TS}} = 2^{-i} \quad \sum_{i=1}^N H_i \xrightarrow{N \rightarrow \infty} 1 \quad (2.3)$$

Thus, the fastest task with periodic time $T_{p1} = 2.T_{TS}$ is assigned 50 % of the complete processing time, task 2 with periodic time $T_{p2} = 4.T_{TS}$ gets assigned 25 %, task 3 gets 12.5 %, ...

From (2.3) it shows, that the total utilization of H aiming with an infinite number of tasks to 100%, thus, there will be no overload situations, if all tasks meet their maximum execution time T_{TS} requirement (easily guaranteed with a preemptive schedule).

$$T_{TS} \geq \max(T_{e1}, T_{e2}, T_{e3}, \dots)$$

With a non-preemptive schedule (cooperative task schedule) maximum execution times T_{ei} have to be determined carefully (WCET determination), such, that each task can complete execution within the basic T_{TS} . If a task gets delayed for some reasons (like high priority interrupts during execution) the complete schedule will get delayed.

Advantages

Binary fixed time-slice scheduling is

- quite suitable with multirate sampling systems (e.g. oversampling, when integer multiples of a common sampling frequency are needed).
- a very simple scheduling without priorities,
- suitable in a non-preemptive realization for even small microcontrollers without OS.
- Time and duration of execution for each task is guaranteed (apart from interrupts)

Disadvantages

- Binary time-slice scheduling is not fully flexible with arbitrary periodic times and execution times
- can turn into very complex software, especially with integrating tasks with execution time larger than T_{TS} .

Lab-Experiment AVR-Timer-Interrupts-(FTS-Scheduling)

```

void FTS_schedule()           // called periodically, with FTS_us
{
    if (FTSLICE(1, FTS_count)) { // ---- 2^1 timeslice ----
        // -----
        TOGGLE(togpin);          // for WCET measurements by scope
        adc_in = analogRead(adcpin);
        TOGGLE(togpin);          // for WCET measurements by scope

    } else if (FTSLICE(8, FTS_count)) { // ---- 2^8 timeslice ----
        // -----
        TOGGLE(outpinA);

    } else if (FTSLICE(9, FTS_count)) { // ---- 2^9 timeslice ----
        // -----
        TOGGLE(outpinB);

    } else if (FTSLICE(10, FTS_count)) { // ---- 2^10 timeslice ----
        // -----
        TOGGLE(ledpin);           // visual heartbeat ~ 1.0 Hz
    }
    FTS_count++; // increment timeslice counter
}

```

Sketch for Arduino Nano (FixedTimeSlice.ino)

Example: 4 periodic tasks (on an AVR µC, from section Digital PID Control with 4 Tasks)

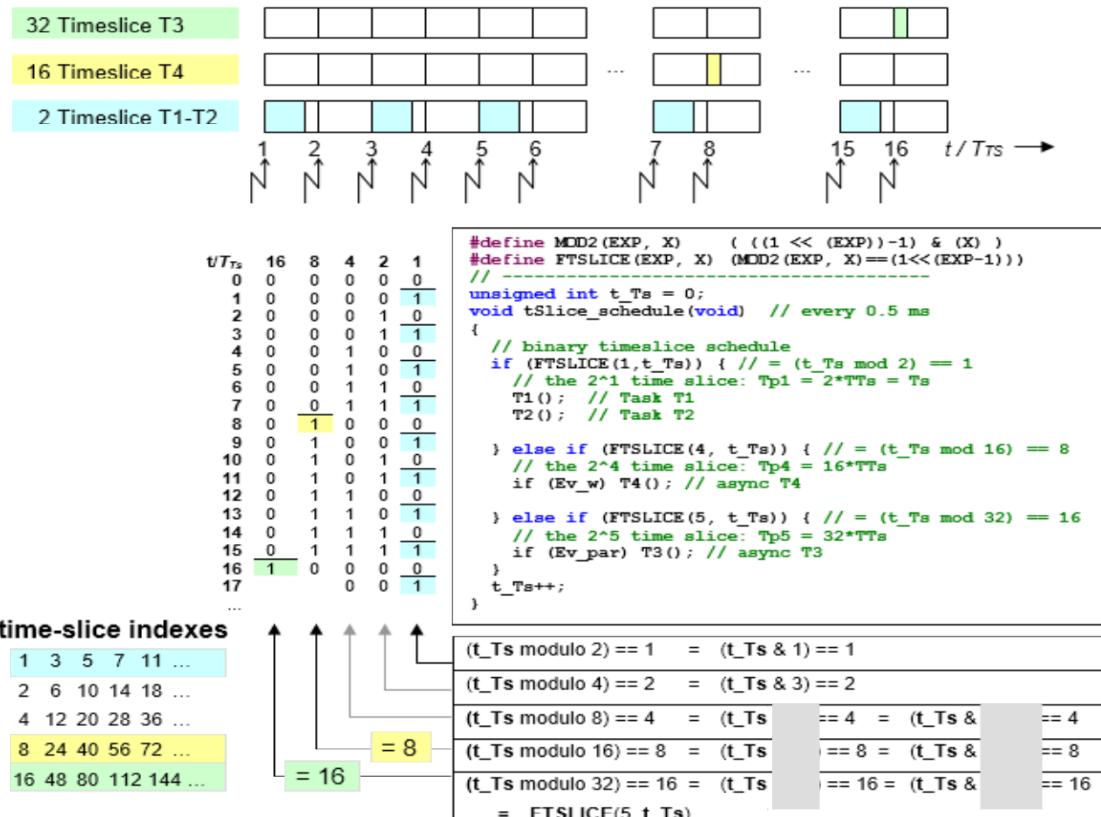
$$T_{p1} = T_{p2} = 1 \text{ ms}, T_{p3} = 16 \text{ ms}, T_{p4} = 8 \text{ ms}$$

$$T_{e1} = 50 \mu\text{s}, T_{e2} = 0.3 \text{ ms}, T_{p3} = 40 \mu\text{s}, T_{p4} = 40 \mu\text{s}$$

$$H_1 = 5\%, H_2 = 30\%, H_3 = 0.1/20 = 0.5\%, \rightarrow H_4 = 0.05/20 = 0.25\% \quad H = 30.75\%$$

Solution with Binary time-slice schedule with 3 Tasks, $T_{TS} = 0.5 \text{ ms}$, $T_{TS} \geq \max(T_{ei}) = \max(0.3, 0.05, 0.005, 0.0025 \text{ ms})$ tasks with the same periodic times run in the same time-slice:

→ T1 und T2 executed successively in the 2 Timeslice $2 \cdot T_{TS} = 1 \text{ ms}$



Proof for collision-free assignment of time-slice indexes (with $k, l \in \mathbb{Z}$ integers)

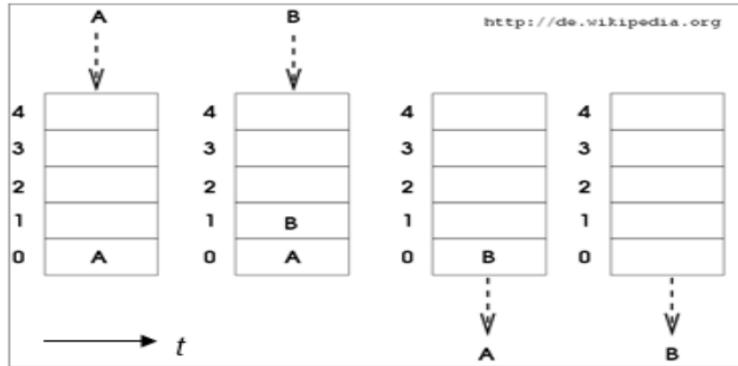
for some period N we get time slices at indexes $n_k = k \bullet N + \frac{N}{2}$ e.g.: 2, 6, 10, 14, ..

thus, for period $2N$ (next time slice) we get $m_l = l \bullet 2N + N$ e.g.: 4, 12, 20, 28, ..

for a collision, we set $n_k \stackrel{!}{=} m_l$. Division by N on both sides shows $k + 0.5 \stackrel{!}{=} 2 \bullet l + 1$, which has no solution with integers k and l , thus the above assignment is without any conflict all time slice indexes (proof by full induction) !

2.1.3 FIFO Scheduling

The name derives from the FIFO (first in, first out) principle of a waiting queue:



where elements put into a queue in a certain order of sequence, are taken out in the same order of sequence. With a **FIFO scheduler**, the tasks are processed in the same order in which they have taken the ready state.

→ An running task is not interrupted, it is a **non-preemptive, dynamic** scheduler.

Advantage: very simple implementation, sometimes used in universal OS

Disadvantage: bad real-time performance, violations of time conditions even at low processing demands (example)

Example: Two tasks with FIFO scheduling

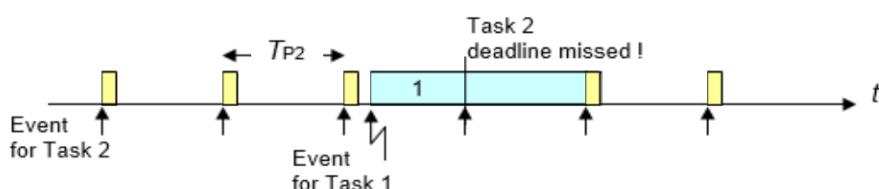
Task 1: $T_{p1} = 150 \text{ ms}$, $T_{e1} = 15 \text{ ms}$ $H_1 = 0.1$

Task 2: $T_{p2} = 10 \text{ ms}$, $T_{e2} = 1 \text{ ms}$ $H_2 = 0.1$

For both periodic tasks there is a time bound (deadline) with the next period. With (2.2)

$$H = H_1 + H_2 = 0.1 + 0.1 = 0.2$$

For 20 % processor demand only, it should be easy to find a schedule for this taskset, but:



This simple example shows that even at a very low processor demand, a FIFO scheduler **cannot guarantee** compliance with the real-time conditions.

Fixed-Priority Scheduling

For fixed-priority scheduling, each task is assigned a fixed priority. The **ready-task** with the highest priority waiting in the scheduler's queue is assigned to the processor.

→ fixed-priority scheduling is **dynamic scheduling with static priorities**. (often used with interrupt processing of microprocessors, preemptive or non-preemptive):

1. Fixed-Priority-Preemptive Scheduling (FPP)

If a **task with higher priority** than the currently running gets into the ready state, then the **current task is interrupted (displaced, preemted)** and the task with the higher priority is assigned to the processor immediately.

2. Fixed-Priority-Non-Preemptive Scheduling (FPN)

If a **task with higher priority** than the currently running gets into the ready state, then the **task is assigned** to the processor only **after the current task** is either **completed** or **blocked**.

Advantage: with FPP, the compliance with the real-time conditions can be guaranteed (unlike to FIFO scheduling), if the priorities were assigned appropriate.

The assignment of priorities to tasks is an important step in the development of a real-time application.

Rate-Monotonic-Scheduling (RMS)

For purely periodic applications, there is a rule, the so-called **rate-monotonic scheduling rule**, which states, that the priority of the tasks to be performed is inversely proportional to their periodic time:

$$PR_i \sim \frac{1}{T_{pi}} \quad (2.4)$$

PR_i Priority task i, T_{pi} Periodic time of task i

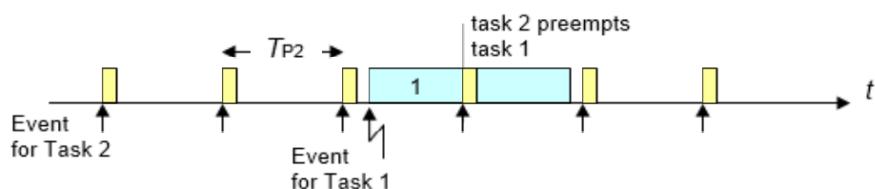
under the (rather idealized) conditions:

- Preemptive scheduling is used
- The periodic times T_{Pi} are constant
- The time limits (deadlines) are equal to the periodic times T_{Pi}
- The execution times are known and constant T_{ei}
- The tasks are independent from each other (can not deadlock)

Example: (same as with FIFO scheduling in section 2.2.6). Two tasks with Fixed-Priority-Preemptive Scheduling (FPP):

Task 1: $T_{p1} = 150$ ms, $T_{e1} = 15$ ms, lower priority by (2.4)

Task 2: $T_{p2} = 10$ ms, $T_{e2} = 1$ ms higher priority by (2.4)



The running task 1 is displaced with an event for task 2, which has higher priority (its state goes from **blocked ready**), so that the deadline of task 2 is not violated (in contrast to FIFO scheduling).

Obviously, **FPP scheduling** with **RMS** provides much better results than FIFO scheduling.

In contrast, with **FPN**, the deadlines of task 2 would be violated !

In general

With periodic tasks with fixed priorities and deadlines equal to their periodic times, rate-monotonic scheduling (RMS) provides for an optimal priority allocation.

This also means that FPP scheduling with RMS always delivers an executable schedule, if one exists. However, the overall CPU utilization may not exceed H_{max} ([1], by Liu)

$$H_{max} = n \cdot (2^{1/n} - 1) = n \cdot (\sqrt[n]{2} - 1) \quad (2.5)$$

n number of tasks

n	H_{max}
1	1.000000
2	0.828427
3	0.779763
4	0.756828
5	0.743492
10	0.717735
20	0.705298
50	0.697974
100	0.695555
1000	0.693387
10000	0.693171

The limit (2.5) can be used to examine the feasibility of a (periodic) taskset and to guarantee the conformance with all time limits. FPP with RMS is very popular because of its simplicity.

However, there are problems at very high utilization (beyond or near H_{max}) and with RMS at the same or nearly the same time periods, delivering same task priorities. Furthermore, there are sometimes difficulties in approaching non-periodic processes by periodic processes with sufficient precision.

Example 1: 3 Tasks with $H_{max} > 0.778$, from [WörnB] :

Task T1: period $T_{p1} = 10$ ms, execution time $T_{e1} = 1$ ms $\rightarrow H_1 = 0.1$

Task T2: period $T_{p2} = 10$ ms, execution time $T_{e2} = 5$ ms $\rightarrow H_2 = 0.5$

Task T3: non-periodic, deadline $= T_{d3} = 15.4$ ms, execution time $T_{e3} = 5.5$ ms

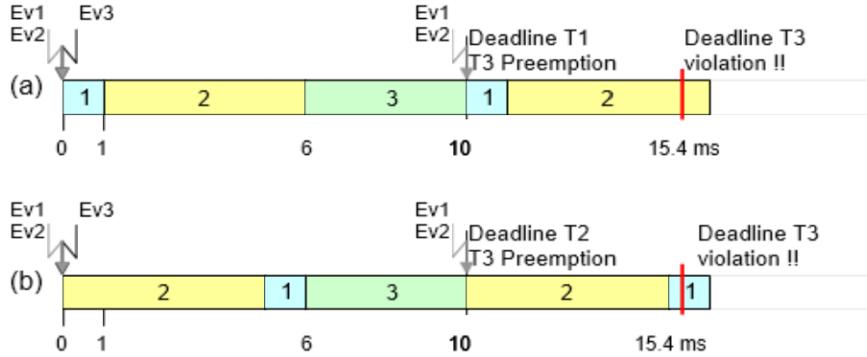
By (2.2) the total CPU utilization $H = 0.1 + 0.5 + 0.357 = 0.957$, almost 100 % ! But with (2.5) $H > H_{max} > 0.78$ deadlines will be violated with **FPP/RMS** with 3 tasks.

FPP/RMS **can't deliver an executable schedule**, as this example shows:

By (2.4) one has 2 choices for assigning the priorities due to the equal periods T_{p1} and T_{p2} :

In both cases there is a violation of the T3 deadline, if all tasks get **ready** the same time at $t = 0$:

Priority assignment (a)	Priority assignment (b)
T1 high priority	T2 high priority
T2 medium priority	T1 medium priority
T3 low priority	T3 low priority



Of course, the violation of T3 deadline in the example above occurs due to the deliberate disregard of the load limit H_{max} , by (??). If one meets the max. load requirement $H < 78\%$ ($n = 3$), there is violation of the T3 deadline.

However, with growing number of tasks the maximum load goes down as low as 70% (from $n > 10$), which is surprisingly low.

Ideas to achieve higher processor loads lead to dynamic assignment of priorities.

Example 2: 3 Tasks with $H_{max} < 0.778$, from [WörnB] :

Task T1: period $T_{p1} = 10$ ms, execution time $T_{e1} = 1$ ms $\rightarrow H_1 = 0.1$

Task T2: period $T_{p2} = 10$ ms, execution time $T_{e2} = 5$ ms $\rightarrow H_2 = 0.5$

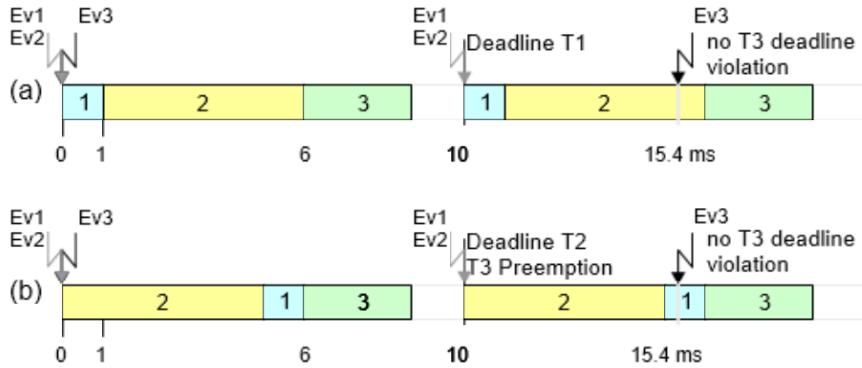
Task T3: non-periodic, period T_{p3} = deadline $T_{d3} = 15.4$ ms, $T_{e3} = 2.62$ ms $\rightarrow H_3 = 0.17$

By (2.2) the total CPU utilization $H = 0.1 + 0.5 + 0.17 = 0.77$ With (2.5) $H < H_{max} = 0.78$ deadlines will be not violated with FPP/RMS with 3 tasks:

By (2.4) one has 2 choices for assigning the priorities due to the equal periods T_{p1} and T_{p2} :

Priority assignment (a)	Priority assignment (b)
T1 high priority	T2 high priority
T2 medium priority	T1 medium priority
T3 low priority	T3 low priority

In both cases there is no violation of the T3 deadline, even if all tasks get **ready** the same time at $t = 0$:



→ If one meets the max. load requirement $H < 78\%$ ($n = 3$), there is violation of the T3 deadline.

However, with 3 tasks the maximum load allowed is as low as 77.8% (for $n = 3$), which is surprisingly low.

Ideas to achieve higher processor loads lead to dynamic assignment of priorities
 → next sections.

Earliest-Deadline-First Scheduling (EDF)

In Earliest-Deadline-First Scheduling (EDF) the Task Processor is granted to that **ready**-task, which is closest to its time limit (deadline). This is achieved by assigning the task priorities according to the vicinity to the individual deadlines.

→ EDF is dynamic scheduling with dynamic priorities, either preemptive or non-preemptive

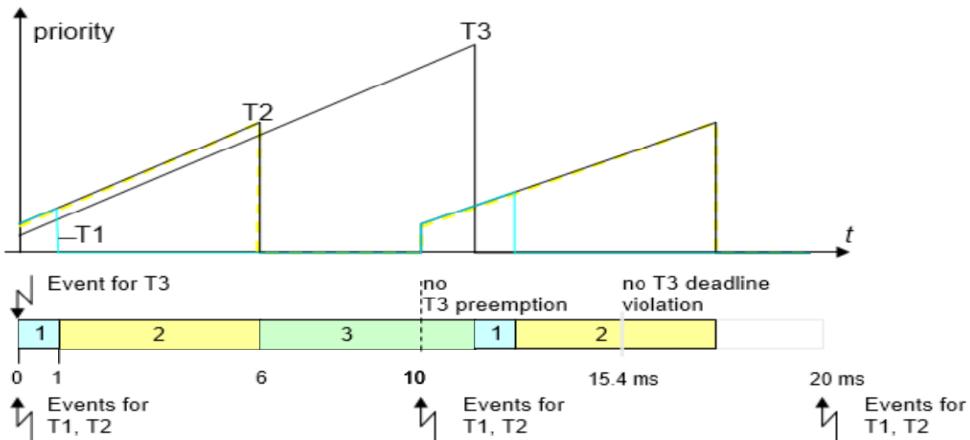
For **preemptive EDF scheduling** a context-switch is made, when a task with earlier time bound gets **ready**. In **non-preemptive EDF scheduling** is the task with the shortest time interval ist executed only, if the currently running task is finished or blocked.

Preemptive EDF scheduling, is used more frequently than non-preemptive EDF scheduling, which is essentially only used when non-interruptible processes are to be managed, such as disk access.

Advantage: With **preemptive EDF scheduling** 100 % CPU utilization can be achieved.

Example with 3 Tasks (same as in section 2.2.7): Task T1: period $T_{p1} = 10$ ms, execution time $T_{e1} = 1$ ms $H_1 = 0.1$ Task T2: period $T_{p2} = 10$ ms, execution time $T_{e2} = 5$ ms $H_2 = 0.5$ Task T3: non-periodic, deadline = $T_{d3} = 15.4$ ms, execution time $T_{e3} = 5.5$ ms(each periodic time is assumed to be deadline).

Again, by (2.2) the total CPU utilization $H = 0.1 + 0.5 + 0.357 = 0.957$, almost 100 % !



(at $t = 0$, T1 and T2 have equal priorities, due to equal deadlines, in such a case, the scheduler decides randomly, e.g. for the task with the lower id)

It can be seen, that with EDF scheduling an executable schedule is found.

Liu [WörnB] shows, that EDF scheduling is an optimal scheduling with

$$H < 100\% \text{ for preemptive EDF scheduling} \quad (2.6)$$

Advantage:

1. as long as the processor utilization is less than 100 % with a uniprocessor system, EDF scheduling delivers an executable schedule and compliance with all time conditions is guaranteed.

Disadvantages:

1. increased computational complexity for the dynamic priorities needed at run time.
2. the sequence of task assignment is difficult to control.
3. the time of execution is difficult to control with fixed time requirements.

2.1.4 Task-Synchronization and Communication

As most of the resources may only be used exclusively, tasks that request these resources must be synchronized for exclusive use, sometimes with regard to some sequence of access.

Thus, the OS provides **means for synchronization** either for

1. mutual exclusion, and
2. cooperation (sequence of access).

Synchronization

The problem of synchronization of tasks arises when these tasks are not independent from each other. Dependencies arise, when common resources are used, then the access needs to be coordinated. Common resources may be

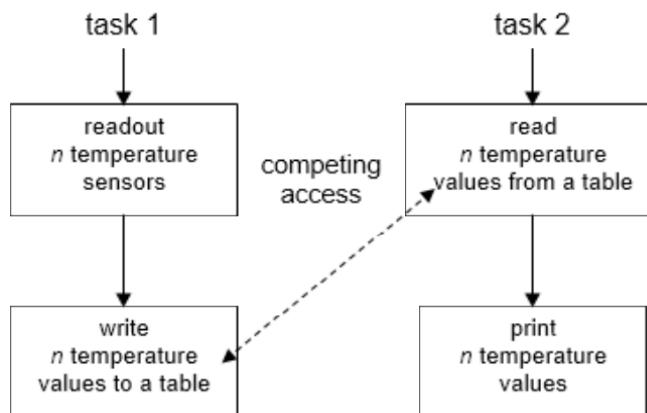
1. **Data:** Multiple tasks read and write access to shared variable, like tables. (without synchronization, uncoordinated access could result in inconsistent data, e.g. when task 1 is reading values of a table, while another task 2 is updating that table partly).
2. **Devices:** Multiple tasks using common devices such as sensors or actuators. (again, coordination is necessary to not to send e.g. contradictional commands of two tasks to a stepper motor drive).
3. **Programs:** Several tasks share common programs, such as device drivers. Competing calls to a device driver must be assured to leave consistent application states.

Example: Two tasks compete for access to a data table:

task 1 is reading several temperature sensors, and stores these values in a table

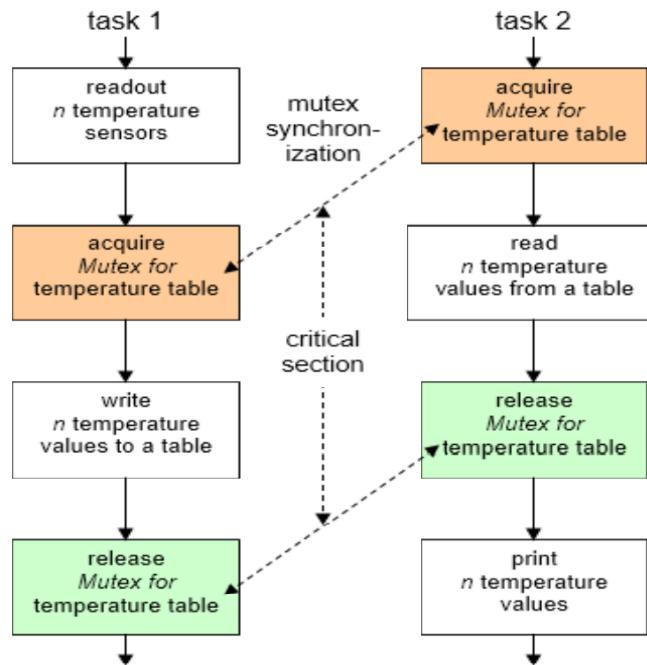
task 2 is reading this table, and prints out the temperature distribution.

Without synchronization, this can lead to erroneous results if, for example task 1, the common table has not yet been fully updated, while task 2 accesses. Then, task 2 gets mixed new and old temperature values, which can lead to undefined states !



There are two basic types of synchronization:

1. The **Mutual Exclusion** or shortly called **Mutex**, ensures that access to a common resource is permitted only to one task at a time.

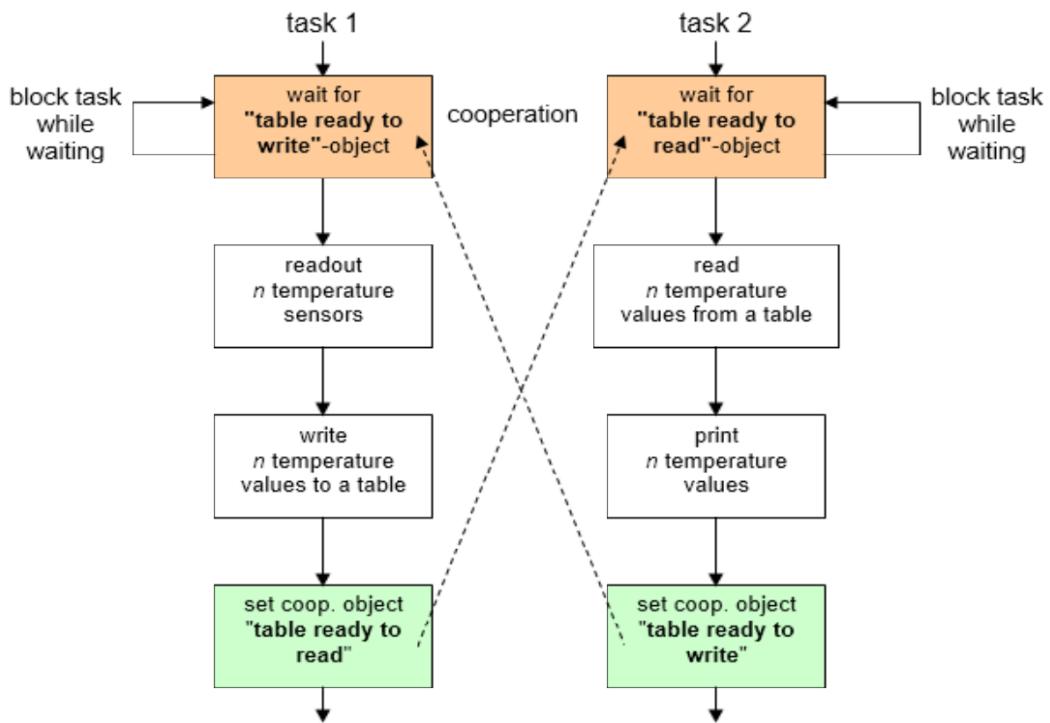


Example: of temperature measurement using the **mutex** synchronization.

For the protection of the common temperature table, a **mutex** is defined, to exclude the possibility that both tasks **access** this resource simultaneously. Before entering the critical section a task tries to acquire the mutex from the OS. If the mutex is already occupied by the other task, the newly accessing task is blocked until the other task **releases** the **mutex** again. The task will **un-block**, acquisition of the **mutex** succeeds, allowing the task to enter the critical section.

The sequence, which task 1 or 2 acquires the **mutex** does not matter.

2. Synchronization, where the order of access to common resources matters is called **cooperation**
 –unlike as with **mutex synchronization**, which doesn't regard the order of task requests.



Both, mutex synchronization and cooperation objects can be realized with semaphores.

2.1.5 Semaphores

A semaphore (from the Greek σύμφρενος = "sign" and φέρειν = "carry") is historically a signal mast or flag signal, in Information Technology it is an object for synchronization of processes.

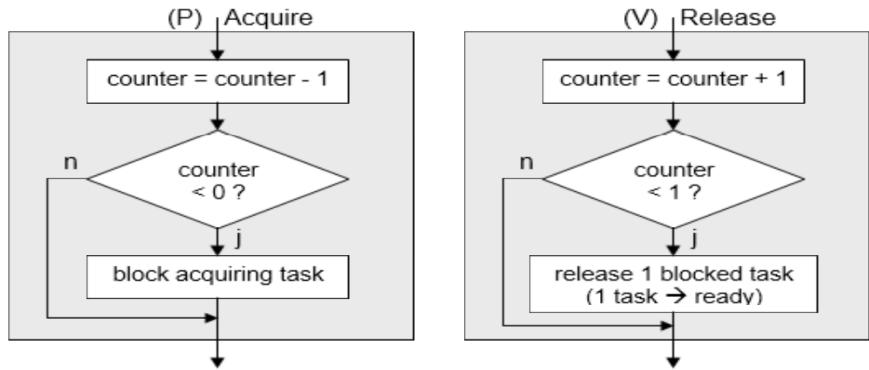
A **semaphore** is basically a counter variable and two non-interruptible **operations**:

1. **Acquire** (also called "**take**", "**signal**", "**lock**", or "**Passieren**", (**P**))
2. **Release** (also called "**give**", "**wait**", "**unlock**" or "**Verlassen**", (**V**))

If a task tries to **acquire** a semaphore, an associated counter variable is decreased. As long as the counter variable has a value less than 0, then the acquiring task is blocked.

Thus, during initialization the positive value of the semaphore states the number of tasks -waiting in a FIFO queue, or in a list using priorities- that may pass the semaphore, and thus enter the critical region protected by the semaphore.

If a task **releases** a semaphore the associated counter variable is increased again. If the value of the counter variable is smaller than 1, tasks trying to acquire the semaphore are blocked, or, if the counter is greater or equal than 1, a waiting task is released from its blocking state the task gets ready. Thus, a negative semaphore counter value indicates the number of tasks that have been prohibited to pass a semaphore. **Binary semaphore** only use **0** and **1** as counter values.

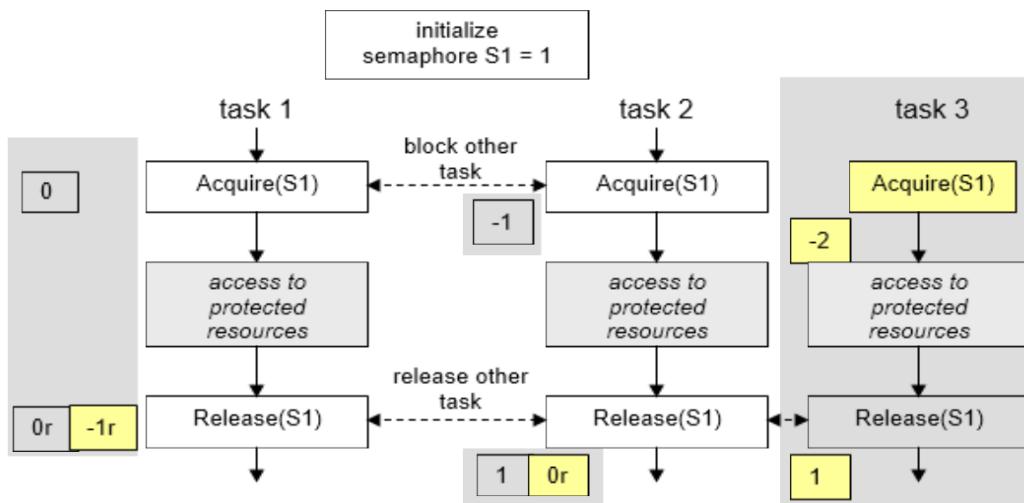


It is of crucial importance that the operations "Passieren" and "Verlassen" (as originally introduced by Dijkstra) are realized atomically, i.e. they cannot be interrupted by any other operation. Only then, a consistent handling of the counter variable is ensured. Semaphore objects are managed by the RT-OS.

Other names for the semaphore operations:

P(sem) = Passieren(sem) = sem.P() = sem.decrement() = sem.wait() = take(sem)_{VXWorks}
V(sem) = Verlassen(sem) = sem.V() = sem.increment() = sem.signal() = give(sem)_{VXWorks}

For **mutex-synchronization** a semaphore is used, with counter initialized with 1. Therefore, only one task can pass, gaining access to the protected resource:



Example: Semaphore as Mutex in FreeRTOS

```

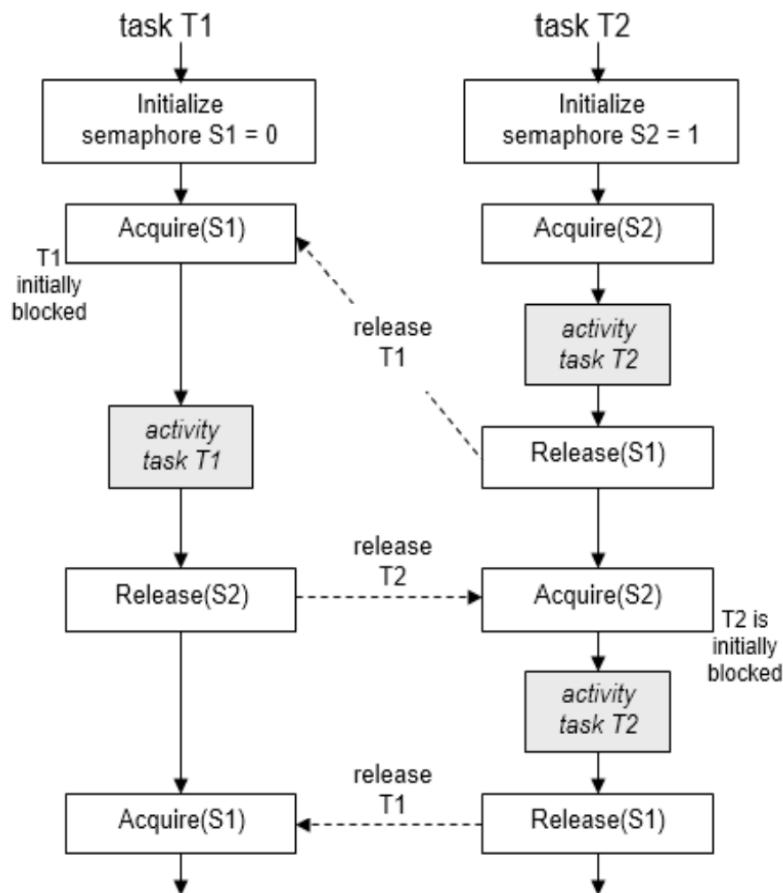
1 xSemaphore semAD; // a global Semaphore
2 int table[16]; // a table of 16 temperatures
3
4 void vAppTask(void * pvParameters)
5 {
6     semAD = vSemaphoreCreateBinary(); // Create a Semaphore, initially 1
7
8     while(1) {
9         ...
10        xSemaphoreTake(semAD, portMAX_DELAY); // ----- sync acquire mutex
11
12        // read temperatures exclusively ...
13        Read_Temperatures(table);
14

```

```

15     xSemaphoreGive(semAD, portMAX_DELAY);      // ----- sync release the mutex
16 }
17 }
18
19 void vPrintTask(void * pvParameters)
20 {
21     while(1) {
22         ...
23         xSemaphoreTake(semAD, portMAX_DELAY);    // ----- sync aquire mutex
24
25         // print the table of temperatures exclusively ...
26         Print_Temperatures(table);
27
28         xSemaphoreGive(semAD, portMAX_DELAY);    // ----- sync realease the mutex !
29     }
30 }
```

A **cooperation synchronization** can be realized by means of two semaphores. Example: sequential order of a task cooperation T2T1T2 with 2 semaphores:



Example: Cooperation in FreeRTOS

```

1 void init()
2 {
3     s1 = vSemaphoreCreateBinary();          // Create a binary Semaphore
4     s2 = vSemaphoreCreateBinary();          // Create a binary Semaphore
5     initSemaphore(s1,0);
6     initSemaphore(s2,1);
7 }
```

```

8
9 void vTask1(void * pvParameters)
10 {
11     while(1) {
12         xSemaphoreTake(s1,portMAX_DELAY); // <---- sync take blocks task !
13         t1_activity();                // main T1 activity
14         xSemaphoreGive(s2,portMAX_DELAY); // <---- sync
15     }
16 }
17
18 void vTask2(void * pvParameters)
19 {
20     while(1) {
21         xSemaphoreTake(s2,portMAX_DELAY); // <---- sync take blocks task !
22         t2_activity();                // main T1 activity
23         xSemaphoreGive(s1,portMAX_DELAY); // <---- sync
24     }
25 }
```

2.1.6 Deadlocks

Synchronization of tasks can lead to a deadlock (*block*, *deadly embrace*), a situation, where the continuation of one or more tasks is blocked permanently. A simple example of a deadlock is an intersection with right-before-left right-of-way rule and four simultaneously arriving vehicles. After the priority rule any vehicle must wait for being right to another one, none of the Vehicles can drive. Only a break of the rule can release the deadlock. In task management, there are two types of deadlocks: deadlocks and livelocks.

In a deadlock tasks waiting on the release of resources for which they block each other. This leads to a standstill of the waiting tasks. The previous example with the traffic intersection describes such a deadlock.

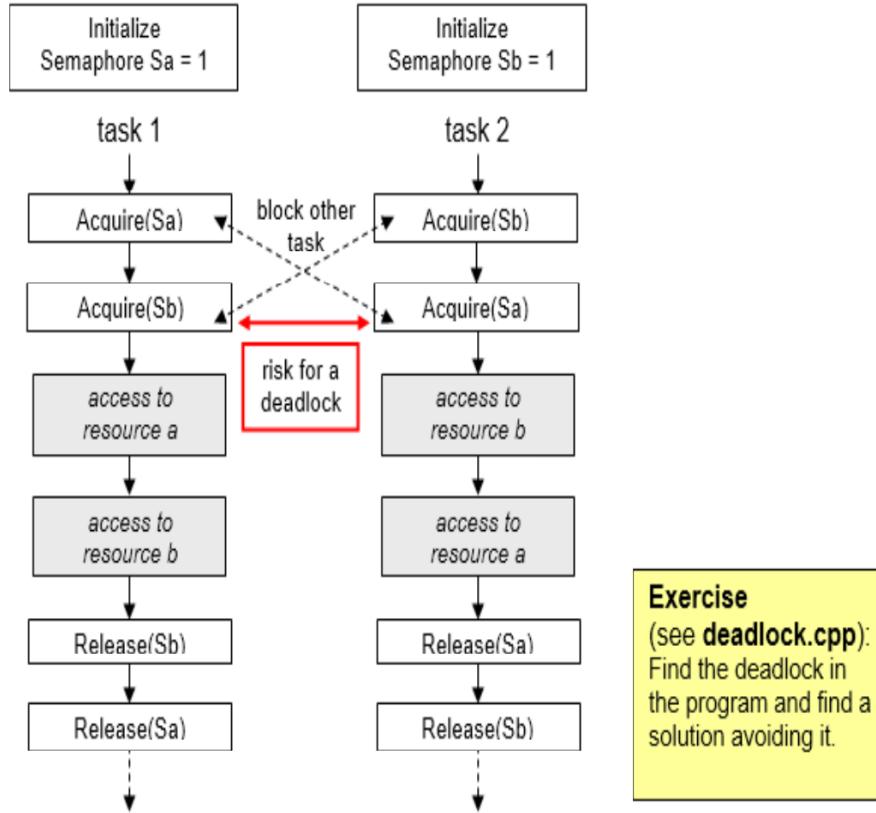
Deadlocks typically occur when multiple resources are to be protected at the same time over cross: In the above program there is a risk for a deadlock, since whenever both tasks are executed nearly parallel in time, both semaphores Sa and Sb are acquired, but never released. Deadlocks are usually realized unintended, i.e. as *software bugs* !

A simple rule to avoid deadlocks:

If several resources are to be protected at the same time, all accessing tasks must acquire / release semaphores in the same order to avoid deadlocks.

Another, although less elegant, method is to remove deadlocks already occurred, by a defined waiting period for any task in blocked state, a so-called *timeout*. If, after the time-out, a task is still in blocked state, that task is reset and all depending resources it occupied are released.

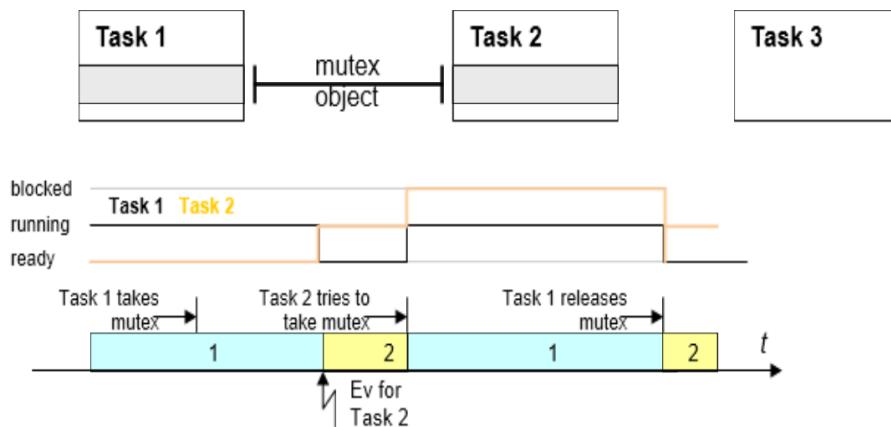
This technique is implemented in some RTOSes, see also → Software Watchdog concept.



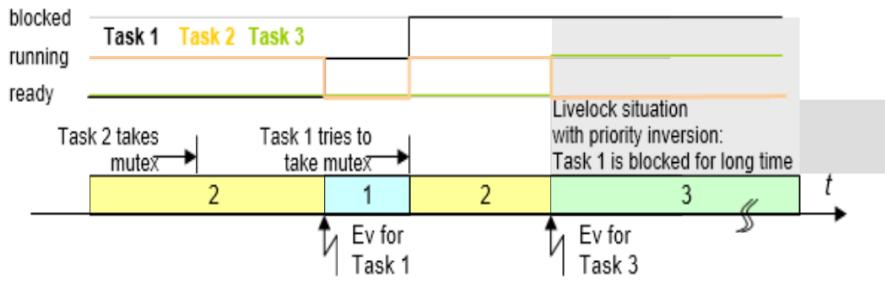
2.1.7 Livelocks

Livelocks (starvation) designate a condition in which a task is constantly inhibited from running by the conspiracy of other tasks. For example, when using a fixed lower-priority task never can get the processor due to the constant activity of higher-priority tasks.

Example FPP, task 1: high priority, task 2: low priority, task 3: medium priority



However, this can also happen in high-priority tasks, provided that a **priority inversion** occurs: the high-priority task 1 is livelocked by the lower priority tasks 2 and 3:



The problem of livelock by priority inversion can be solved by the technique of **priority inheritance** [WörnB] in the example above: task 2 inherits the high priority of task 1 (both wait for the same mutex), such that task 2 running in the critical section cannot be interrupted by a medium priority task 3.

A priority inversion condition occurs when a high-priority task must wait for synchronization objects of a lower-priority task (with **fixed priority scheduling**). This was the problem of the 1997 Mars mission "Sojourner" (running VxWorks), which was finally rescued by a watchdog [WörnB].

2.1.8 Task Communication

Synchronization and communication tasks are very closely related. **Synchronization** can be thought of as **communication without information**. On the other hand, communication can be used for synchronization.

An example could be waiting for an order. With the order, information is transmitted, and by the time the order is sent, synchronization is possible (the job can start actually).

There are two basic variants of the task communication:

1. **Shared memory:**

The data exchange is via a shared memory.

The synchronization at what time read or write access occurred is done by a semaphore.

2. **Messages (Events):** The data exchange and synchronization is done via sending messages.

In general, the communication via shared memory is faster than communication via messages. Therefore, this variant is preferred used in real-time operating systems.

In spatially **distributed systems**, this is not possible because there is no physically shared memory. Here communication must take place via messages (→ message queues).

2.1.9 Events

With a mutex synchronization the access of competing tasks to a jointly used resource was enabled.

In contrast, for cooperating tasks time (and order) of execution has to be synchronized. This can be done by semaphores, but also with a simpler concept, an **event**.

A task can wait on an **event** (event wait), which is set by another task. Once set, the waiting tasks gets released from its blocking state.

Example:

Each of 3 tasks (task A, task B, task C) acquires a measurement value, which shall be processed by a 4th task W. If A, B or C has acquired a measurement value, this is written to a data memory ("datenspeicher"), which can store exactly one value. Task W retrieves its input from that data memory.

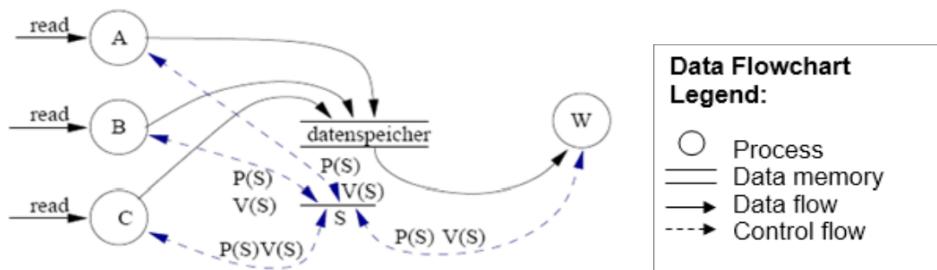


Figure 2.1.1: Data flow of an acquisition of measurement values: synchronization without events

Since the data memory (datenspeicher) is existing only once, it is a shared resource, and access to it is a critical section, which can be protected by a semaphore S

```

1 // Task A,B und C
2 ...
3 while( TRUE ) {
4     read(kanal_A, buffer, sizeof(buffer));
5     P(S); // enter critical section
6     write( datenspeicher, buffer, sizeof(buffer));
7     V(S); // leave critical section
8 }
9 ...
10 }
```

```

1 // Task W
2 ...
3 while( TRUE ) {
4     P(S); // enter critical section
5     read(datenspeicher, buffer, sizeof(buffer));
6     V(S); // leave critical section
7     WorkOnData( buffer );
8 }
```

Problems with synchronization without events:

the **data memory** can be overwritten, without being read before by W

W can read the **data memory** repeatedly.

Therefore, it is not sufficient to protect the shared resources by using a semaphore, yet another synchronization means is necessary: an **event**.

Definition:

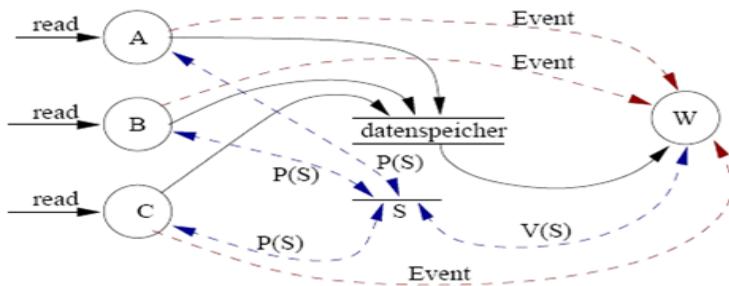
An **event** is a synchronization object, which allows computing processes to give free the processor (by going to the **block**-state), up to a certain condition is met.

Basic operations of the event are:

- Wait for an event, and
- Send an event.

An **event** can get "lost", if an event is sent when no process waits on it !

Since events are not stored -unlike semaphores-, events are used often in combination with a semaphore.



Data flow of an acquisition of measurement values: synchronization with events

In order to grant access to the data for the tasks A, B or C only, after task W has emptied the data memory, the operation "enter critical section" and "leave critical section" is distributed to different tasks, similar to cooperation synchronization with semaphores.

Each of the tasks A, B and C enters the critical section (by calling the "P(S)"-operation), but never "leaves the critical section" by itself, this release is now done by task W (by calling the "V(S)"-operation). Task W must now be synchronized, so that it fetches the data from memory only, if these are available. Thus, Task A, B or C send an event on which task W is waiting:

```

1 // Task A,B und C
2 ...
3 read(kanal_A, buffer, sizeof(buffer));
4 P(S); // enter critical section
5 write( datenspeicher, buffer, sizeof(buffer) );
6 SendEvent( Task_W );
7 ...

```

```

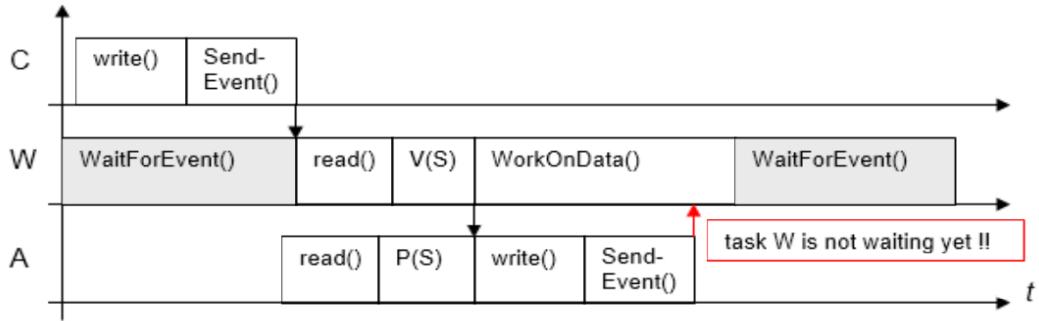
1 // Task W
2 while (1) {
3     WaitForEvent( Task_W );
4     read( datenspeicher, buffer, sizeof(buffer) );
5     V(S); // leave critical section after read
6     WorkOnData( buffer );
7 }

```

The critical section is thus defined from the time which an acquiring task A,B, or C writes a value into the data memory, until task W has read the value out of the memory.

Problem: Events can get lost, if sent when no task is waiting

- task C causes task W to get running, starting with read()



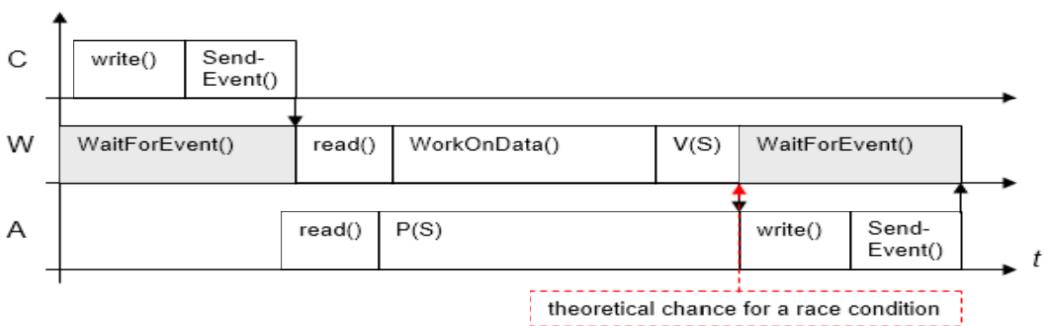
- During read() in task W, task A has read new data
- task A is blocked by P(S), before W could release the semaphore
- task W releases V(S)
- task A continues with write() and SendEvent()
- the event gets lost, cause task W is busy WorkingOnData(), not waiting: ***critical race condition !!***
- data loss A.

Solution: Move the WorkOnData() call into the critical section:

```

1 // Task W
2 while (1) {
3     WaitForEvent( Task_W );
4     read( datenspeicher, buffer, sizeof(buffer) );
5     WorkOnData( buffer );
6     V(S); // leave critical section after work
7 }
```

Now, with this solution, the conflict is solved, since immediately after the critical section is left, task W can wait for events there is no more critical race condition:



Since there is a theoretical (although very small) chance, **Posix** defines an atomar, non-interruptable operation combining the semaphore release and waiting on an event:



```

1 // Task W
2 P(S); // since WaitForEventAndV releases a semaphore
3     // this must be locked (acquired) first
4 while(1) {
5     read( datenspeicher, buffer, sizeof(buffer) );
6     WorkOnData( buffer );
7     // V(S); leave critical section after work, done above
8     WaitForEventAndV(S); // Release semaphore and wait
9 }
```

```

1 // Task A,B und C
2 ...
3 read(kanal_A, buffer, sizeof(buffer) );
4 P(S); // enter critical section
5 write( datenspeicher, buffer, sizeof(buffer) );
6 SendEvent( Task_W );
7 ...
```

Posix Functions:

By these functions a critical section is entered, and left

P(S): int **pthread_mutex_lock** (pthread_mutex_t * *mutex*);

V(S): int **pthread_mutex_unlock**(pthread_mutex_t * *mutex*);

WaitForEventAndV(S):

int **pthread_cond_wait**(pthread_cond_t * *cond* , pthread_mutex_t * *mutex*);

Alternative Solution: increase Memory, use Message Queues as a standard solution !

2.1.10 Signals

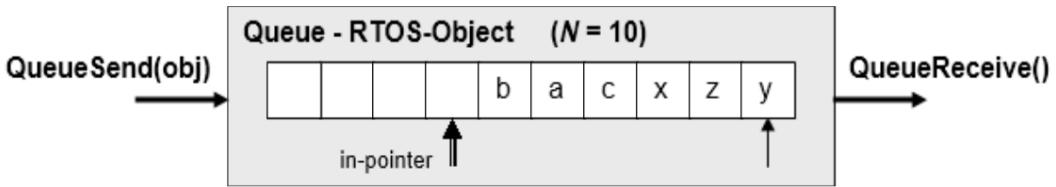
Signal are very similar to events. Signals can be processed asynchronously to the program (call of a signal-handler routine), with events the processing is synchronous (**wait_for_event()** - functions as part of the program).

- Signals are often used in UNIX programs.
- Signals they can be seen as software interrupts at the application level.

2.1.11 Messages Queues, Mailboxes

Message queues provide an asynchronous communication protocol, which means that the sender and receiver of the message must not interact simultaneously. Messages will be placed on a queue, where they are stored until the recipient retrieves one or more by a FIFO mechanism. The maximum amount of data a single message is usually limited.

A message queue without a restriction for the number of messages is called **mailbox**.



Example: FreeRTOS - Queue (www.freertos.org)

```

1 #include <FreeRTOS/queue.h>
2 ...
3 struct AMessage
4 {
5     portCHAR ucMessageID;
6     portCHAR ucData[ 20 ];
7 }
8
9 xQueueHandle xQueue1;
10
11 void vATask(void *pvParameters)
12 {
13     // Create a queue capable of containing 10 pointers to AMessage structures.
14     // These should be passed by pointer as they contain a lot of data.
15
16     xQueue1 = xQueueCreate( 10, sizeof( struct AMessage * ) );
17 }
18
19 ...
20
21 // ----- send from any task -----
22 {
23     // Send a pointer to a struct AMessage object.  Don't block if the
24     // queue is already full.
25     struct AMessage *pxMessage;
26
27     pxMessage = \& xMessage;
28     xQueueSend( xQueue1, (void *) \&pxMessage, (portTickType) 0 );
29 }
30 ...
31
32 // ----- receive with any task -----
33
34 {
35     // Receive a message on the created queue.  Block for 10 ticks if a
36     // message is not immediately available.
37     struct AMessage *pxMessage;
38     if( xQueueReceive(xQueue1, \&(pxRxedMessage), (portTickType) 10 ) )
39         // pcRxedMessage now points to the struct AMessage variable posted
40 }

```

The **QueueReceive()** command places the calling task into the ***blocked (waiting)*** state if the message queue is empty. If not, it returns the object to which the read (out-) pointer points to, and the read (out-) pointer set to the next message object.

The **QueueSend()** command places the calling task into the **blocked (waiting)** state only, if the mes-

sage queue is full. If not, the object is being stored at the position the write (in-) pointer references, and the pointer is set to the next free space in which the next transmitted (posted) message object is stored.

In many RTOS, a maximum blocking time can be specified (as with FreeRTOS).

Lab-experiment "Message Queue with ARM LPC4357 µC (FreeRTOS)"

2.1.12 Socket Interface

The most important interface for inter-process communication on different computers connected via Ethernet, is the **socket interface**.

By using sockets, data can be exchanged between processes, which are located on different hosts (distributed system).

The socket interface provides access to TCP/IP (connection-oriented) and UDP/IP (connectionless) protocols for communications.

Once a socket connection between two processes has been established, the data can be exchanged with some basic system functions (open, read, write, close ...), the so-called

Sockets Service Primitives

Primitive	Meaning	
SOCKET	Create a new communication end point	
BIND	Attach a local address to a socket	
LISTEN	Announce willingness to accept connections; give queue size	blocking
ACCEPT	Block the caller until a connection attempt arrives	
CONNECT	Actively attempt to establish a connection	
SEND	Send some data over the connection	
RECEIVE	Receive some data from the connection	blocking
CLOSE	Release the connection	

→ RTP ProgC Repetition Workspace: "socket_client, socket_server"

Python Sockets Programming:

The server connects to a client and echoes its message:

```
server:> socket server.py
Connected by ('192.168.0.3', 4118)

server:>
```

```
client:> socket client.py
Received 'Hello World'

client:>
```

(A) Server script

```
1 # http://docs.python.org/library/socket.html#socket-example
2
3 import socket
4
5 # Echo server program
6 import socket
7
8 HOST = ''                      # Symbolic name meaning all available interfaces
9 PORT = 50007                     # Arbitrary non-privileged port
10 s = socket.socket(socket.AF_INET, socket.SOCK_STREAM)
11 s.bind((HOST, PORT))
12 s.listen(1)
13 conn, addr = s.accept()
14 print 'Connected by', addr
15 while 1:
16     data = conn.recv(1024)
17     if not data: break
18     conn.send(data)
19 conn.close()
```

First, the server executes LISTEN (line 12), and it remains blocked until a client connects. Then a client task executes CONNECT to establish a connection with the server (line 9). It must specify an IP address (the server's) and port number (must be common to both).

(B) Client script

```
1 # http://docs.python.org/library/socket.html#socket-example
2
3 # Echo client program
4 import socket
5
6 HOST = '192.168.0.10'          # The remote host
7 PORT = 50007                   # The same port as used by the server
8 s = socket.socket(socket.AF_INET, socket.SOCK_STREAM)
9 s.connect((HOST, PORT))
10 s.send('Hello, world')
11 data = s.recv(1024)
12 s.close()
13 print 'Received', repr(data)
```

The OS then sends an IP packet to the opposite computer (peer). The client process is blocked until the server responds.

If a confirmation is received, the client and server have established a connection and the connect() - or listen() function are released from blocking, and data can be exchanged until the end of the connection (close()).

2.2 Memory Management

The main task of memory management is essentially providing access to the memory resource consisting of a hierarchy of cache, main memory (RAM), peripheral memory or NVRAM (non-volatile RAM, such as EEPROM) by

- the allocation of memory to the tasks,
- the coordination of accesses to shared memory areas,
- the protection of the storage areas of individual tasks
- the displacement of memory areas

One can distinguish between different forms of memory allocation:

1. Static memory allocation:

The allocation of memory is made to a task before it is put into the *ready* state. The memory allocation does not change at runtime.

2. Dynamic memory allocation:

The allocation of memory to a task is done at runtime and may change at any time.

3. Non-displacing memory allocation:

Allocated memory must not be withdrawn from a task at runtime.

4. Displacing memory allocation:

Allocated memory may be removed from a task at runtime, the memory is paged to peripheral memory (e.g. a swap-file).

Embedded systems mainly use **static memory allocation**, since the dynamic allocation of memory is hardly predictable (and thus real-time compliant). Execution times of **malloc()**, **free()**, **new()**, **delete()**, .. can't be guaranteed to be smaller than a certain limit availability.

The usage of an **MMU** (Memory Management Unit), which is quite common with many CPUs (x86, ARM Cortex A Series, ..), allows processors to deal with a virtual address space. Thus, memory access from the process is strictly separated from the OS and from other processes (= safety). However, with translation of the virtual addresses into physical addresses long delays can occur, which cannot be determined in advance! This is a contradiction to the required determinism with hard real-time requirements ! Thus, microcontrollers for applications having hard real-time requirements often do **NOT** have an MMU and use direct physical addressing instead (AVR, ARM Cortex M Series, ..)



2.3 Input/Output Management (I/O)

Along with task- and memory-management the I/O-management is the 3rd important component of a real-time operating system, especially with embedded systems, where I/O tasks for sensors and actors have hard real-time requirements.

In addition to that, synchronous and asynchronous serial interfaces, and networking subsystems require high performance I/O-management.

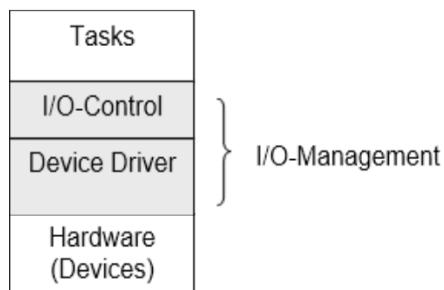
I/O Basics

The tasks of the I/O-management can be divided into two groups:

- assigning and sharing of devices for the tasks
- use of assigned and shared devices by the tasks

The connected devices differ greatly in speed and data format. The OS has therefore the task, to abstract from implementation details and provide a **simple, uniform and transparent interface**.

This can be achieved by a multi-layer architecture of the I/O management:



The main 2 layers of I/O-management:

1. I/O-Control:

This layer is above the layer of the device driver and interfaces with the tasks. It is hardware independent, i.e. implementation and operation is no longer depending on the type of connected devices. It takes care for data transport, data format, data buffering, and messaging. If a certain device needs to be exchanged, its device driver needs to be exchanged too, but not the I/O control. The I/O control abstracts from the specific devices and provides a **simple, uniform and transparent interface** to the **tasks**.

2. Device Driver:

The device driver is a layer next to the hardware, i.e. its realization depends on the connected devices. It takes into account all device-specific properties, including direct communication with the devices. It produces all necessary control signals, evaluates the status signals and handles the device's interrupts. The device driver monitors all transfers, and possibly do some error handling, e.g. evaluate parity bits and initiate the repetition of a failed transmission.

Example: I/O-management of a hard disk drive:

Each drive has a unique device driver, that takes into account their specific characteristics such as speed, number of heads, cylinders and sectors. At the level of I/O-control, each disk is treated equally,

for the task addressing of data is done by a block number, abstracted from the device-specific parameters:

```
FILE *fp = fopen("c:\test.bin"); fseek(fp, ...), fwrite(fp, ...), fclose(fp);
```

Typical interface functions provided by the I/O control layer

Function	Description
Create	Creates a virtual instance of an I/O device
Destroy	Deletes a virtual instance of an I/O device
Open	Prepares an I/O device for use
Close	Communicates to the device that its services are no longer required, which typically initiates device-specific cleanup operations
Read	Reads data from an I/O device
Write	Writes data into an I/O device
Ioctl	Issues control commands to the I/O device (I/O control)

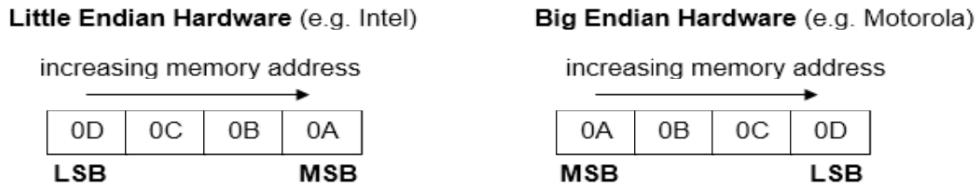
The main functions of the I/O control layer are:

1. **Symbolic Names:** devices addressing by symbolic addresses, or names.
2. **Handling of I/O-Requests:** queuing, buffering I/O-requests queues.
3. **Assignment of Devices:** assignment of the device (shared resource) to tasks dynamically or statically, using priorities for competing requests.
4. **Synchronization:** I/O-requests affect a tasks state, (e.g. waiting **blocked** state), interface between I/O-control to task-management.
5. **Device Protection:** prevent unauthorized access by unauthorized tasks - protection against deadlocks and livelocks due to competing access- resets after programmable timeouts (protect against software failure).
6. **Communication with the device drivers:** forwarded executable device requests to the device driver, handle feedback, e.g., status or error codes
7. **Buffering:** synchronization of different speeds of task-management and I/O-devices by temporary storage in buffer memory, handle buffer state events to the task/device
8. **Unique Data Format:** abstract from the device's physical data format, e.g. **endianness**.

The boundary between I/O control and device drivers is floating and system dependent. For efficiency reasons with some RTOSes, functionality is moved into the device driver, which increases the effort for adapting new devices.

Endianness

The storage of the hexadecimal number $0x0A0B0C0D = 168496141$ results in a different byte order in memory, dependent of the microcontroller or peripheral hardware:



1. With **little-endian** hardware the least significant byte (**LSB**) is stored with the **lowest memory (byte) address**,
2. With **big-endian** hardware the most significant byte (**MSB**) is stored with the **lowest memory (byte) address**

Most RTOS's provide functionality to abstract from **endianness** in the device driver. Example:

htonl, htons, ntohs, ntohs - convert values between host and network byte order

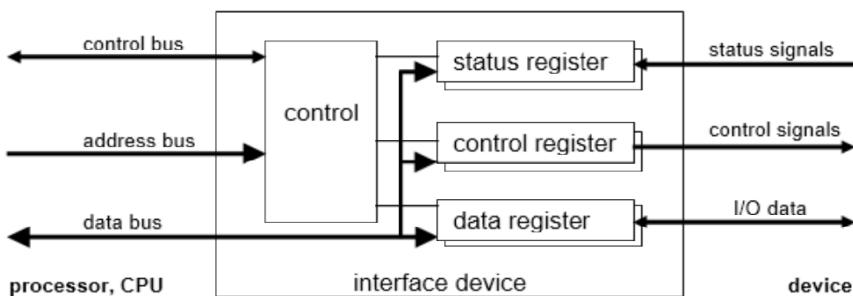
```
#include <arpa/inet.h>
uint32_t htonl(uint32_t hostlong);
uint16_t htons(uint16_t hostshort);
uint32_t ntohl(uint32_t netlong);
uint16_t ntohs(uint16_t netshort);
```

The **htonl()** function converts the unsigned integer *hostlong* from host byte order to network byte order.
The **htons()** function converts the unsigned short integer *hostshort* from host byte order to network byte order.
The **ntohl()** function converts the unsigned integer *netlong* from network byte order to host byte order.
The **ntohs()** function converts the unsigned short integer *netshort* from network byte order to host byte order.

On the i386 the host byte order is Least Significant Byte first, whereas the network byte order, as used on the Internet, is Most Significant Byte first.

2.3.1 I/O Synchronization

The communication between peripheral devices and the processor is performed by interface devices (**interface units**). These **interface devices** also define the base for synchronization between the processor (tasks), and devices operating at different speeds using certain registers:



The typical components of an interface device are:

1. **Data register(s)** to read/write the data to be received/transmitted.
2. **Control register(s)** for configuring the device (e.g. operating mode, speed of serial ports, e.g. UART).

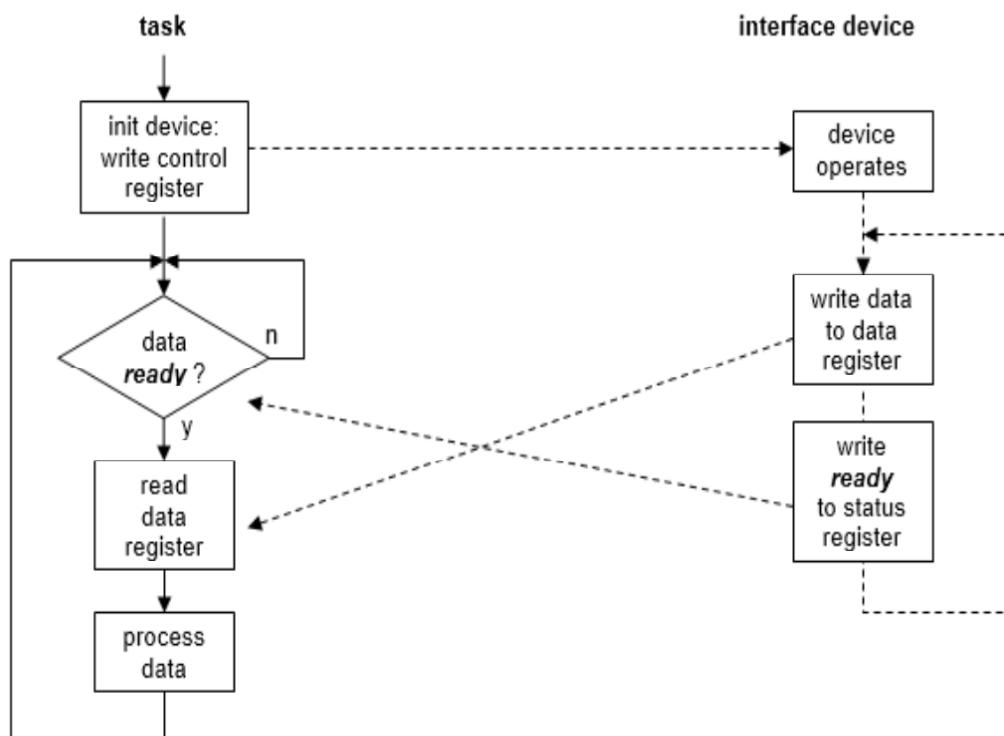
3. **Status register(s)** reflecting the devices current state (e.g. sent data, received data bits, e.g. ADC).

A running task can synchronize (for changing data) with the interface device either by

1. Polling
2. Interrupts

Polling

With the simplest synchronization method –polling- a task reads the status register continuously in a loop:



After initialization (by setting the control register) the task is waiting in an active loop on a status flag –which is a particular bit within the status register of the device- indicating that data is available in the data register

→ the task reads the data from the data register.

Advantages of polling:

Simple synchronization type

Disadvantages of polling:

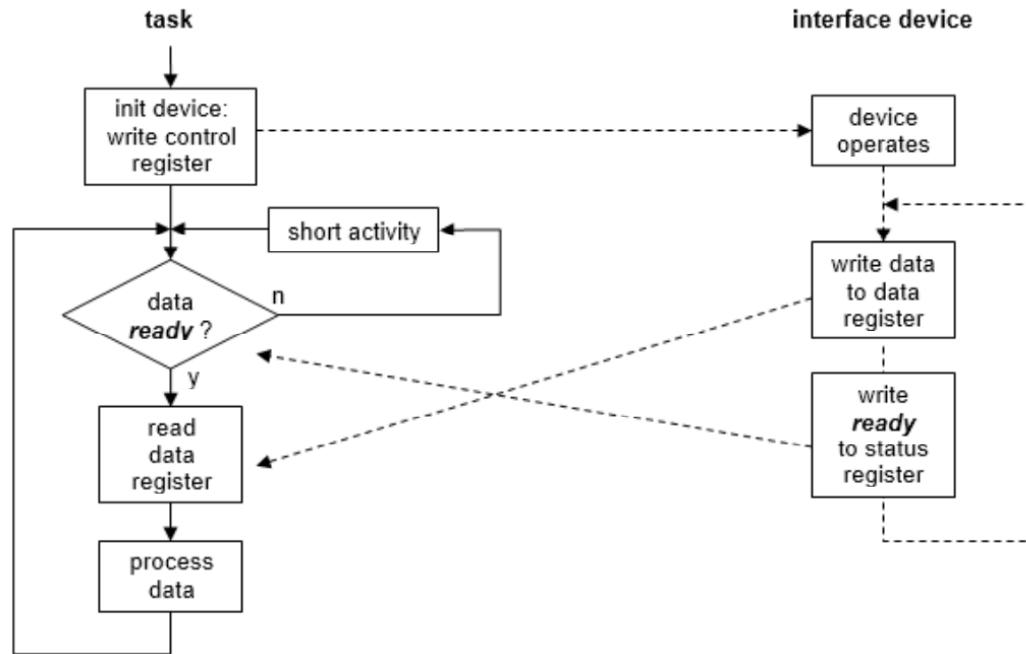
Consumption of processor time for the active loop

Slow response times when multiple devices within a task are polled simultaneously.

Conclusion: Overall, polling is usually not favorable, due to inefficient CPU usage.

Busy Polling

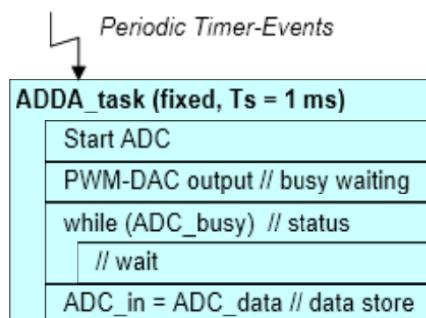
With busy polling, the properties of simple polling are improved with performing some activity (typically short) within the waiting loop:



A disadvantage is, however, the response time slows down. Longer activities must be split into smaller steps.

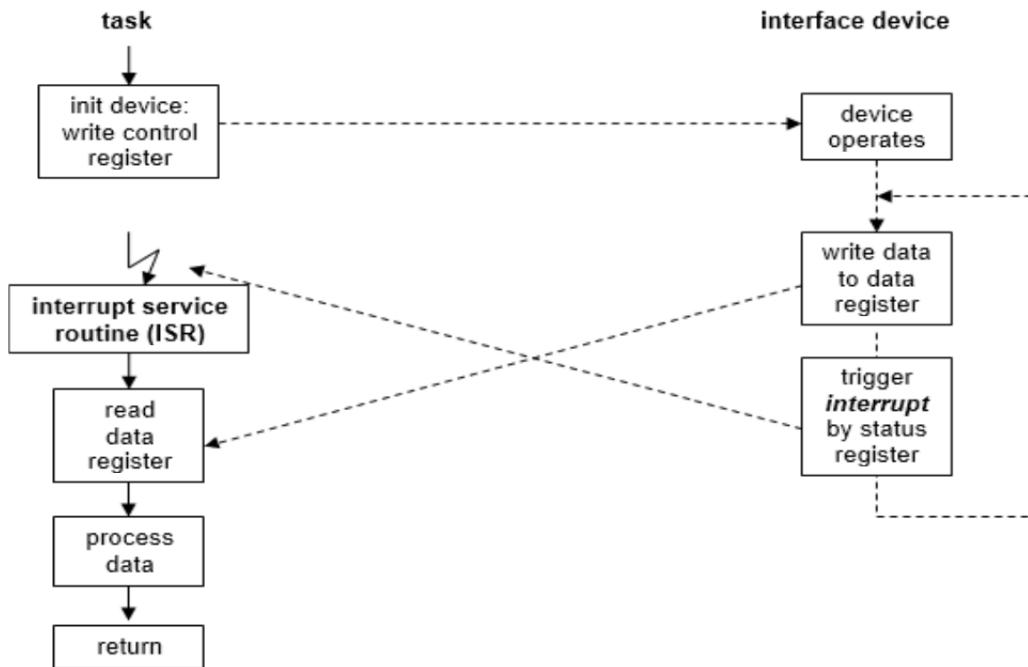
Example (sampling system (PWM-Control Experiment in Real-Time-Lab)):

- Start an AD converter process with known duration,
- Offset / Gain correction of previously read values,
- Wait for the ADC to get ready,
- Readout of the ADC data register.



Interrupts

With synchronization by interrupts the interface executes an interrupt request with the processing CPU, which forces a currently running task to be interrupted in favor of a defined interrupt service routine **ISR**:



In the interrupt service routine (ISR), the data can be read from or written to the device.

Advantage:

Processing time is consumed only, when data needs to be transferred

Disadvantage:

Activating the ISR causes an overhead as the current processor state must be saved for the old task to resume after return of the ISR (equivalent to preemption with task-scheduling).

To reduce the effect of overhead, larger amounts of data are transmitted in blocks, usually in conjunction with direct memory access (DMA).

The technique of synchronization with interrupts is very well suited for real-time applications and thus interrupts are widely used with microcontrollers with both synchronous and asynchronous real-time programming.

2.3.2 Interrupts and Task-Scheduling

Interrupt handling in real-time operating systems raises the question, how these interruptions integrate with the task scheduling. On current processor CPUs a hardware interrupt causes the automatic invocation of an interrupt service routine (ISR).

The real-time operating system then has two options:

1. Interruption of the Task-Oriented Processing:

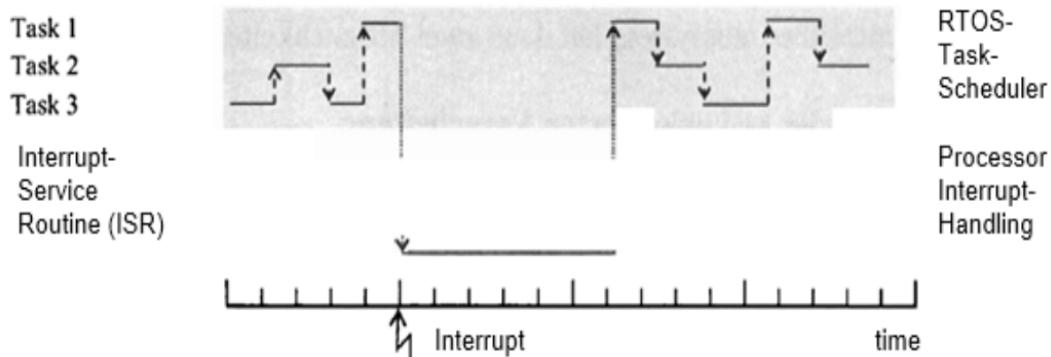
- With an interrupt the currently running task is interrupted. The interrupt service routine (ISR) is executed immediately, after return the interrupted task continues.

Advantage:

- Very fast response to external events, because the interrupts are handled directly by the CPU hardware.

Disadvantages:

- The regular task-scheduling (eg EDF, FPP, TSS ...) is bypassed by direct handling of the processor hardware.
- The processor uses its own priority based preemption (FPP), might conflicting with real-time scheduling



- Regardless of the scheduling strategy used by real-time scheduling events are treated with FPP scheduling, which can not guarantee 100% processor utilization (see section 2.2.7).
- This kind of direct processing of interrupts with interruption of the task-oriented processing is used for RTOS's with less performant microcontrollers, better solution: integration of interrupts with task-scheduling.- The total available processor utilization for task-scheduled processing is reduced from originally 100 % !

2. Integration of Interrupts with the Task-Oriented Processing:

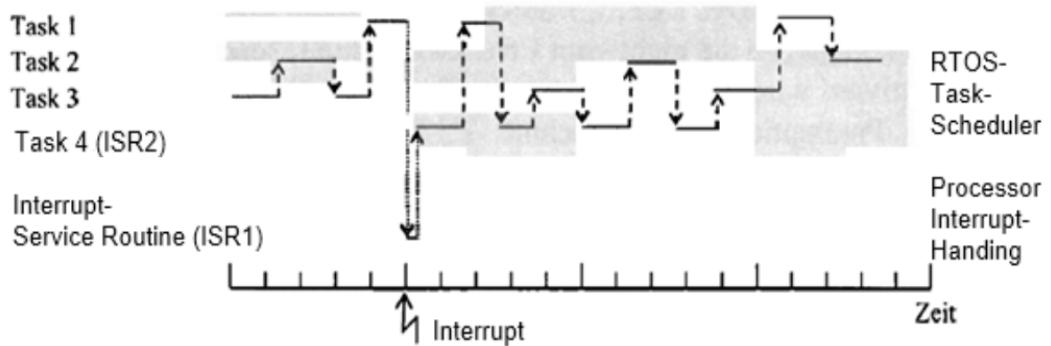
There is a separate task, integrated with task-scheduling, which is responsible for interrupt processing. Thus, whenever the processor gets a hardware interrupt, the ISR starts that separate task, responsible for interrupts.

Advantage:

- Interrupt handling is fully integrated with task-scheduling, there are no more two schedulers simultaneously, which might conflicting.
- An interrupt has the same scheduling policy as all other tasks. (e.g. an interrupt can have lower priority than the actually running task, such that it is not interrupted, with solution 1, this is not possible).

Disadvantage:

- The reaction to events is slowed, because the interrupt handling is not activated directly by hardware any more, but by the task-scheduler.



The integration of interrupts into the task-oriented processing is suitable for real-time operating systems using more powerful micro-controllers.

2.4 Specialized RTOS

2.4.1 Overview of some RTOS

Product	QNX	Posix.4	RT-Linux	VxWorks	OS-9	FreeRTOS	Windows CE	OSEK-OS
Producer	QNX Software	Posix	GPL-GNU Public License	Wind River	Microware	freertos.org Open Source	Microsoft	osek.org
Type	DS, RTOS	RTOS, ExtOS	ExtOS	RTOS	RTOS	MEBS	RTOS	MinOS
Target-System	IA32	IA32, IA64 PowerPC	IA32, PowerPC, ARM	IA32, PowerPC, 68k, div. µC	IA32, PowerPC, 68k	div. µC	IA32, PowerPC, MIPS ARM	div. µC
Languages	C, C++	C, C++, Java, Ada	C, C++, Java	C, C++, Java	C, C++, Java	C, C++	C, C++, Java	C
File-System	Unix, Windows	Unix	Unix, Windows	Unix, Windows	Windows	-	Windows	-
GUI	X-Win	X-Win	X-Win	X-Win	X-Win	-	Windows	-
Network	TCP/IP	TCP/IP, UDP/IP	TCP/IP, UDP/IP	TCP/IP	TCP/IP	TCP/IP	TCP/IP	-
Fieldbus	-	-	CAN	CAN, Profibus ?	CAN, Profibus, Interbus S	CAN	-	CAN, LIN, FlexRay
Scheduling	FPP,FPN, Timeslice, FIFO	FPP,FPN, Timeslice, User-def.	FPP,FPN, Timeslice	FPP,FPN, Timeslice	FPP,FPN, Timeslice	FPP,FPN, Timeslice	FPP,FPN	FPP,FPN, Timeslice

2.4.2 Classification of Real-Time Operating Systems

For classifying RTOSes there can be made a separation into five basic types [1]:

1. Minimal Real-Time Operating System (MinOS)

- Simple RTOS for microcontroller with limited memory resources.
- RTOS is a library, to be linked with the application.
- Simple I/O-functions No memory management, physical addressing only (no protected mode).
- threads only (lightweight processes)

2. Controller System (CS)

- Minimal real-time operating system
 - + a file system
 - + better error handling.

3. Dedicated System (DS)

- Controller System
 - + memory management.
 - + protected mode.

4. Standard OS Extension (ExtOS)

- Standard OS (Windows, Linux, ..)
+ some RTOS features

5. Universal Real-Time OS (RTOS)

- Fully featured RTOS.
+ features of standard OS.

2.4.3 Selection Criteria

A first selection criterion is the classification described in 2.6, which is set by the scope and functionality of a real-time operating system (RTOS).

Other criteria selecting an RTOS are:

1. Development and Target Environment

- Are there well-proven tools and libraries for both development and target ?
- Is the desired programming language supported (C, C++, ..) ?
- Smooth, reliable and fast communication process between the target system and development system (re-flash cycle) ?
- What are the debug options and debugging tools on the target system (E.g., source-level debugging)?

2. Modularity and Core Size

- This criterion refers to the configuration and the memory requirements.
- Can the system be configured flexibly to the application ?
- Can it be restricted to the necessary components ?
- What is the achievable size, application + libraries ?

3. Adaptability

- Is the RTOS adaptable for a particular microcontroller/peripherals ?
- How much effort is that adaption ?
- What field-bus systems or networks are supported ?

4. General Features

- What is the user interface of the operating system (GUI, cmd, None) ?
- Which libraries are part of the operating system ? (e.g. math, graphical, ...)
- What other tools are offered, (e.g. for version management, data management, ..)

5. Performance

(usually a major criterion for selection).

- How large is the maximum number of tasks ? (small automation app: 10 tasks, complex app: > 100 tasks)
- What task-scheduling methods are available ?

- What is the number of different priority levels ?
- What is the time for a context-switch ?
- What is the latency time with responding to interrupts ?
- For which class of real-time application the OS is designed ? (e.g. can hard or firm real-time requirements be met ?)

2.4.4 VxWorks

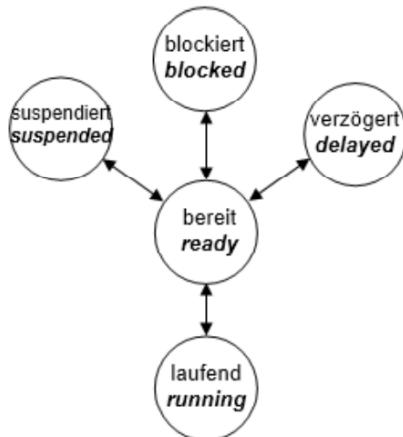
VxWorks, by Wind River Company [2004], is a general purpose real-time operating system. In its basic form, it supports no virtual memory (→ advanced product VxVMI).

VxWorks uses heavyweight processes as tasks. However, task-switch was optimized to enable fast task switching.

The context VxWorks stores in a task-control block (TCB) includes:

1. the program counter of the task,
2. the contents of processor registers and floating-point,
3. the stack of the dynamic variables,
4. the associated I / O devices,
5. the signal handler and
6. various debugging information

The VxWorks state model for the tasks differs not much from the general model from section 2.1.3:



1. **running** the task is executed (run).
2. **ready** (ready) the task is ready to execute.
3. **blocked** (pending) the task is waiting for the release of a resource and is blocked.
4. **delay** (delayed) the task is stopped for a certain period of time.
5. **suspended** (suspend) the task was suspended. This is intended for debugging purposes only.

By default, VxWorks uses FPP scheduling. Each task is assigned an initial priority. This priority can be adjusted during runtime (function **taskPrioritySet()**).

VxWorks defines 256 priority levels, with

1. level **0** corresponding to the **highest**, and
2. level **255** to the **lowest priority**.

The VxWorks task scheduler can be turned on and off explicitly by two functions (**taskLock()** and **taskUnlock()**). The switch will prevent, that the current task is preempted by a higher priority task realization of non-preemptive scheduling (FPN).

However, a task can still be interrupted by hardware interrupts. Deactivation of the task-scheduler is called **preemption locking**. It can be used for the protection of (short) critical sections.

VxWorks also supports Time Slice Scheduling for tasks with equal priority.

Priority inversion

With priority based scheduling the priority inversion problem can arise, which became popular with the mars pathfinder mission "Sojourner", 1997):

Solution: Priority Inheritance [1].

The priority-inheritance protocol assures that a **task that holds a resource executes at the priority of the highest-priority task blocked** on that **resource**. In VxWorks the priority-inheritance protocol can be activated for a mutual-exclusion semaphore with the option **SEM_INVERSION_SAFE** enabled.

For communication between tasks VxWorks provides the following mechanisms:

1. shared memory,
2. signals
3. semaphore,
4. message pipes,
5. sockets to communicate with other computers via network, and
6. non-local procedure calls (Remote Procedure Calls) to other computers over the network.

With VxWorks it is also possible to use libraries or program routines for several tasks simultaneously (shared code) **reentrancy techniques**

2.4.5 FreeRTOS

FreeRTOS is an Open-Source real-time operating system for Embedded Systems. FreeRTOS was ported to a large number of different microcontrollers with different performance:



- Microcontrollers with ARM7-Architektur
- Microcontrollers of the ARM Cortex-M family (M0 . . . M4)
- Altera Nios II Softcore Processor
- Atmel AVR and Atmel AVR32
- Freescale Semiconductor HCS12-Familie and Coldfire V2
- Xilinx MicroBlaze and PowerPC PPC405
- Texas Instruments MSP430
- Microchip Technology PIC18, PIC24, dsPIC, PIC32
- Renesas H8/S SuperH
- Fujitsu MB91460 32bit and MB96340 16bit
- NEC V850ES 32bit and 78K0R 16bit
- x86-Arcitecture
- ...

Features

To ensure good maintainability, FreeRTOS is developed mostly in C, only a few functions are implemented in assembler. The scheduler can be configured for preemptive and cooperative operation. The OS (since version 4) supports two different task classes: "real" tasks and coroutines, the latter are recommended where little space is available.

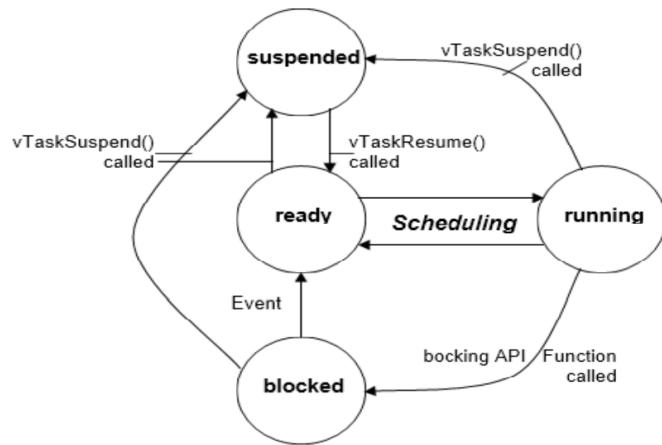
At www.FreeRTOS.org a very extensive documentation to **FreeRTOS**, tutorials and implementation examples realized on various microcontrollers.

SafeRTOS is a derivate of **FreeRTOS**, designed for safety-critical applications by IEC 61508. **SafeRTOS** is cerified by **TÜV Süd** up to SIL level 3.

Example: AD-DA Task of a Sampling Controller

ADDA Task (periodic time 10 cycles)
while (1) // forever
vTaskDelayUntil(&xLastWakeTime, 10); // wait for next cycle
Start_ADC() // Sample & Hold
while (ADC_busy)
// wait
xQueueSend(xQueue, ADC_data) // ADC_data → Queue

2.4.6 FreeRTOS Task-Model



FreeRTOS API-Funktionen (Selection)

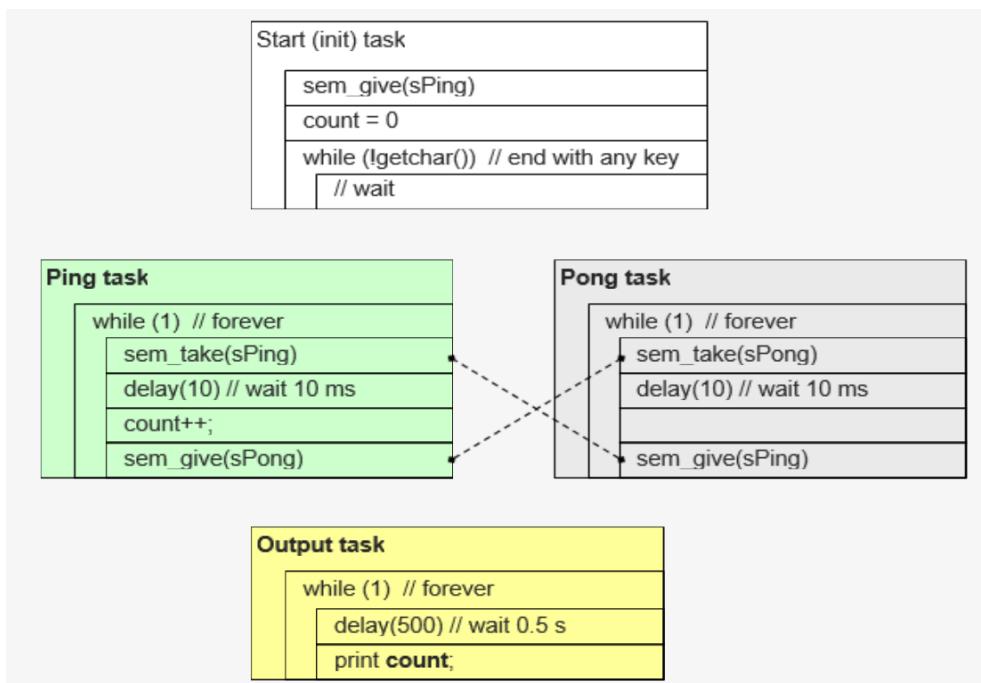
xTaskCreate()	Create a task
vTaskStartScheduler()	Start Scheduler
vTaskSuspend()	
vTaskResume()	
xTaskResumeFromISR()	Resume a suspended task by an ISR
vTaskDelay()	<i>blocking</i>
vTaskDelayUntil()	<i>suitable for exact real-time requirements !</i> <i>blocking</i>
vSemaphoreCreateBinary()	Create a binary Semaphore
xSemaphoreCreateCounting()	
xSemaphoreCreateMutex()	Create Mutex Objekt
xSemaphoreGive()	give (release) a semaphore, V() operation
xSemaphoreTake()	take (acquire) a semaphore, P()-operation <i>blocking</i>
xQueueCreate()	
xQueueSend()	
xQueueReceive()	<i>blocking</i>
xTaskGetTickCount()	Ticks since the Scheduler was started (WCET)

FreeRTOS

- Offers a "Basic Priority Inheritance" mechanism for mutex and semaphore objects for avoidance of **Livelocks**.
- various student projects use FreeRTOS on ARM boards (or similar) at the Real-Time Programming department of the Hochschule Ravensburg-Weingarten → template PingPong

Lab Experiment PingPong

Task diagram for 4 parallel running tasks (3 + 1 init):



```
1 // -----
2 /// \file      PingPong.c
3 /// \brief      RTP-Lab intro.
4 /// \author     Wolfgang Schulter
5 /// \license    for educational purposes only, no warranty, see license.txt
6 /// \date       14.01.2013 ws: initial version
7 // -----
8
9 #include "PingPong.h"          // PingPong module header
10
11 xSemaphoreHandle semPing;      // semaphore Ping
12 xSemaphoreHandle semPong;      // semaphore Pong
13
14 uint16_t count = 0;           // count variable incremented by ping
15
16 #define TICK_RATE_KHZ (configTICK_RATE_HZ/1000)
17 uint16_t delay1 = 10*TICK_RATE_KHZ; // number of ticks <=> 10 ms delay as
18   default
19 uint16_t delay2 = 500*TICK_RATE_KHZ; // number of ticks <=> 500 ms delay as
   default
20 uint8_t do_printf = 0;          // print task: printf cout variable to stdout, if
   1
```

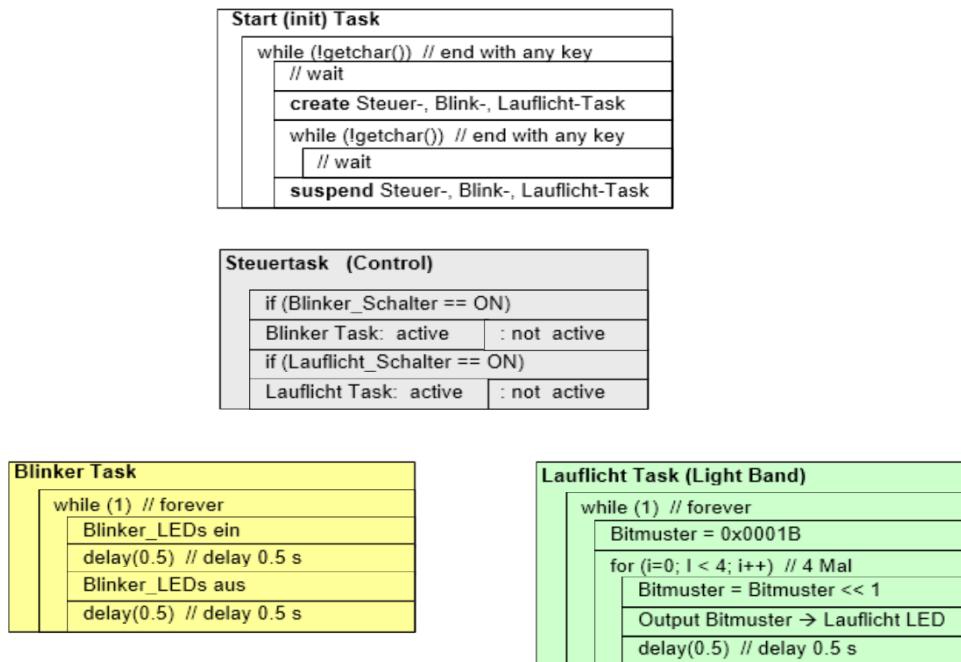
```

20 // = classic mode off by default (we have the GLCD)
21
22 // -----
23 void init_PingPong()
24 {
25     vSemaphoreCreateBinary(semPing);      // <---- init to 1
26     vSemaphoreCreateBinary(semPong);      // <---- init to 1
27     xSemaphoreTake(semPong, portMAX_DELAY); // set to 0
28 }
29
30 // -----
31 void vPingTask(void * pvParameters)
32 {
33     wcet_init(&BWCET_PING); // <---- wcet init
34
35     while(1) {
36
37         xSemaphoreTake(semPing, portMAX_DELAY); // <---- block forever until ...
38         wcet_t1(&BWCET_PING); // <---- wcet measure
39
40         vTaskDelay(delay1);
41         xSemaphoreGive(semPong); // <---- unblock pong task
42         count++;
43
44         wcet_t2(&BWCET_PING); // <---- wcet measure
45     }
46 }
47
48 // -----
49 void vPongTask(void * pvParameters)
50 {
51     wcet_init(&BWCET_PONG); // <---- wcet init
52
53     while(1) {
54
55         xSemaphoreTake(semPong, portMAX_DELAY); // <---- block forever until ...
56         wcet_t1(&BWCET_PONG); // <---- wcet measure
57
58         vTaskDelay(delay1);
59         xSemaphoreGive(semPing); // <---- unblock ping task
60
61         wcet_t2(&BWCET_PONG); // <---- wcet measure
62     }
63 }
64
65 // -----
66 void _print()
67 {
68     _dbg[0] = count; // glcd debug variable _dbg[0]
69
70     if (do_printf) { // since 0.5: classic mode is off by default
71         sprintf(_db, "%d ", count); DB; // print to stdout
72     }
73 }
74
75 // -----
76 void vPrintTask(void * pvParameters)
77 {
78     wcet_init(&BWCET_PRINT); // <---- wcet init
79 }
```

```

80 while(1) {
81     vTaskDelay(delay2);    // wait
82     wcet_t1(&BWCET_PRINT); // ----- wcet measure
83
84     _print();           // GLCD update, print
85     count = 0;          // reset count variable
86
87     LED_TOGGLE(1);      // Toggle LED1
88
89     wcet_t2(&BWCET_PRINT); // ----- wcet measure
90 }
91 }
```

Lab Experiment Multitasking



Lab-Experiment **Multitasking**: The following tasks run (quasi) simultaneously:

1. Start:
TaskGeneration of **Control**-, **Blinker**- and **Lauflicht**-Task. Suspend all tasks after receiving a char from the keyboard.
2. Steuer_Task:
Suspend and resume **Blinker**- and **Lauflicht**-Task depending on the switch position of the switches "**Blinker_Schalter**" and "**Lauflicht_Schalter**".
3. Blinker_Task:
Switch on/off the LEDs in LED_Feld_1 with a periodic time 500 ms.
4. Lauflicht_Task:
Shift-through an active LED from right to left each 500 ms.

2.4.7 OSEK (AUTOSAR OS)

OSEK means "Open systems and interfaces for automotive electronics", it stands for an industrial standard. OSEK is a trademark of Continental AG (up to 2007 from Siemens).

The 1993 committee consists of developers several car manufacturers, their suppliers and software services. Founding members were BMW, Daimler-Benz, Opel, Volkswagen, Bosch, Siemens and the Institute for Industrial Information Technology of the University of Karlsruhe (TH).

The work of the OSEK committee is continued since 2003 under the **AUTOSAR** project www.autosar.org.

OSEK Task Management

The task-scheduler of the OSEK OS distinguishes between

- **Basic Tasks**

run continuously until they are completed, or, until the OS switches to another task, or, when an interrupt occurs.

- **Extended Tasks**

can have a waiting state, by a call of **WaitEvent()** allowing the processor to be released for other tasks

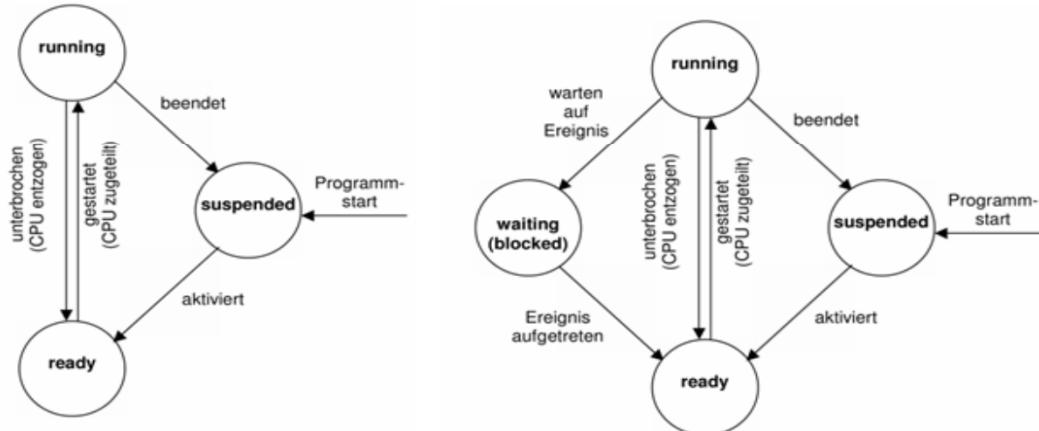


Figure 2.4.1: Task states of Basic Tasks (left) and an Extended Task (right)

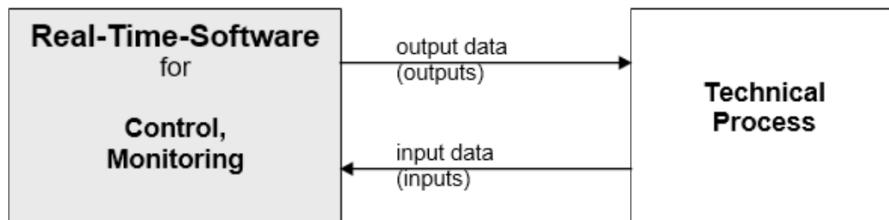
Extended tasks can change into the state of **waiting (blocked)**, if they have the processor, but can not continue because they have to wait for an event (e.g. the controller sends a CAN message and wait for the confirmation).

Chapter 3

Real-Time Programming

3.1 Real-Time Programming

Real-time programming is about the acquisition and the processing of process variables (see section 1.1.1) of a technical process by means of digital programmable processors.

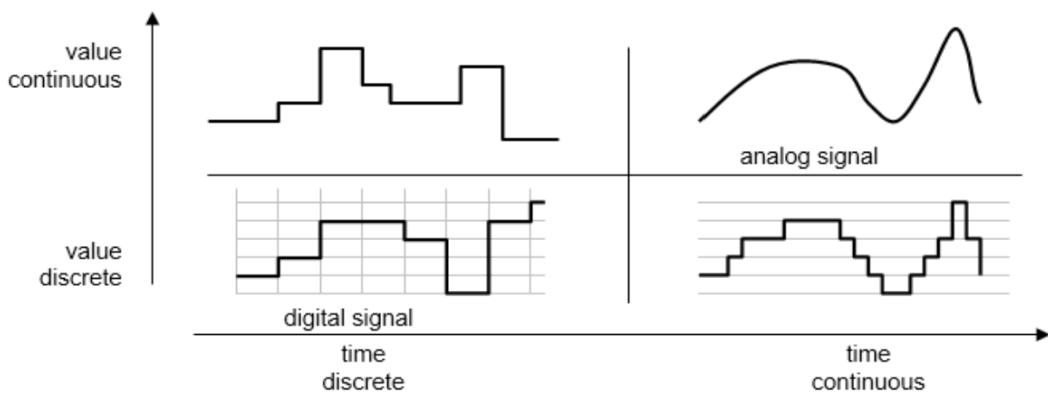


For the acquisition of physical quantities, there are **sensors**, which convert a physical quantity to be measured into an electrical signal (= a function of time).

Analog signals, are **continuous in value** (real valued) and they can be **continuous in time**, or **time discrete**.

Digital signals are **discrete in value**, and either **time discrete** or **continuous in time**.

Since **computers** can **process time discrete** and **value discrete signals** only, analog **sensor input signals** have to be converted into digital (**time- and value discrete**) signals.

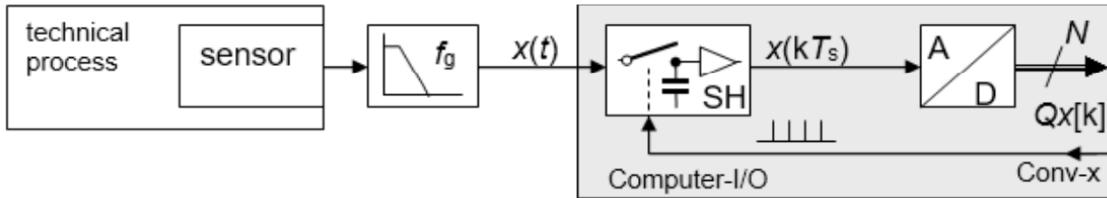


Since a process can be affected by actor (actuator), the **digital, time discrete** control signal, produced by a real-time computer program, must be converted back into an analog signal.

3.1.1 Signal Conversion

An analog sensor signal $x(t)$ (time- and value continuous) is first made time discrete by a sample-and-hold element. The sampling times are given from the I/O system of the computer. Analog signals are sampled at a constant sampling frequency f_s with sampling period $T_s = 1 / f_s$.

It is usually ensured that the signal $x(t)$ has no higher frequency components some limit frequency $f_g \leq 0.5 \cdot f_s$. This is indicated by the analog low-pass filter with cutoff frequency f_g sampling theorem see section 3.1.1.



The amplitudes of the signal $x(k \cdot T_s)$ are still value continuous (real valued), but values change only at discrete times $k \cdot T_s$ (= **sampling**).

In a second step, the time discrete signal $x(k \cdot T_s)$ is converted into a value discrete signal using an ADC (see section 4.5), i.e. the real value x is approximated by an integer number Qx , out of a sequence $(0 \dots N-1)$, such that

$$x \approx Qx \cdot c_x \text{ with } c_x \text{ as some conversion factor} \rightarrow (3.1).$$

After that $Qx[k]$ is a digital signal (time- and value discrete) which can be stored and processes in the computer. With increasing time, the index k increases, e.g. the sequence $\{ Qx[k-2], Qx[k-1], Qx[k] \}$ is the sequence of the last two recent and the actual sampling value. Example:

$$\begin{array}{cccc} \{ & Qx[k-2], & 9 & 5 & -1 \\ & Qx[k-1], & 5 & -1 & 2 \\ & Qx[k] & -1 & 2 & 7 \\ \} & k=0 & k=1 & k=2 & \dots \end{array}$$

a sequence $Qx[k]$ as actual sampling value and with $Qx[k-1]$ as the previous and $Qx[k-2]$ as the pre-previous sample.

In the computer, the value discrete sequence $Qx[k]$ can be identified with the value continuous, time discrete sequence $x[k]$ and thus the analog signal $x(t)$.

The quantization error $e = x - Qx \cdot c_x$ with the sampling process can be minimized, since high resolution AD- and DA-converters are available
 $\rightarrow e$ can be usually neglected.

3.1.2 Digital Sampling Systems

Digital sampling systems are suitable for processing analog signals $x(t)$ with discrete values as a time discrete sequence $x[k] = x(k \cdot T_s)$.

The original analog signal $x(t)$ can be uniquely reconstructed from the time discrete sequence $x[k]$, i.e. the original analog signal has a unique representation by the discrete sequence $x[k]$:

Sampling Theorem of Shannon / Kotelnikov

An analog signal sampled at constant frequency f_s can be reconstructed from the discrete-time sequence of samples (up to a finite delay) if the bandwidth B is not greater than half the sampling frequency, thus $B < 0.5 f_s$.

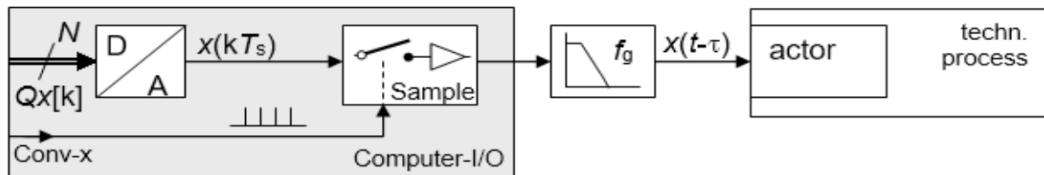
This is the fundamental theorem of digital signal processing.

If the bandwidth condition is violated, i.e. the bandwidth is greater than half the sampling frequency, the analog signal can not be uniquely reconstructed from the sampling, due to irreparable signal distortion, known as **aliasing**.

A digital sampling system, sampling at frequency f_s must ensure that an analog signal is bandlimited before sampling an upper frequency limit $f_g \leq 0.5 f_s$. $0.5 f_s$ is also called **Nyquist frequency** [3]. Thus, the sequence clearly determines the associated analog signal, which couldn't be processed by a digital computer directly!

(The proof is with another theorem of Fourier transform, stating that periodic sampling of a continuous time signal results in a periodic spectrum).

From the sampling theorem follows how the analog signal $x(t)$ can be reconstructed from the discrete time sequence $x[k]$ by lowpass filtering of the discrete time signal:



The reconstruction is done by a periodic pulse train, sampling the DA converted, thus value discrete signal, which is finally (analog) lowpass filtered at cutoff frequency $f_g = 0.5 f_s$

The output of the so-called "reconstruction lowpass" is $x(t)$, which is apart from a certain constant time delay the analog signal $x(t)$!

Real-Time Conditions for Sampling Systems

It is of great importance for signal quality, that strict periodicity of the sampling period. Any deviations from strict periodicity acts as signal interference, which may have some unpredictable consequences with closed loop sampling controllers (e.g. instable control).

A typical requirement for max. **jitter** (average deviation) of sampling period is 0.1% to 1%.

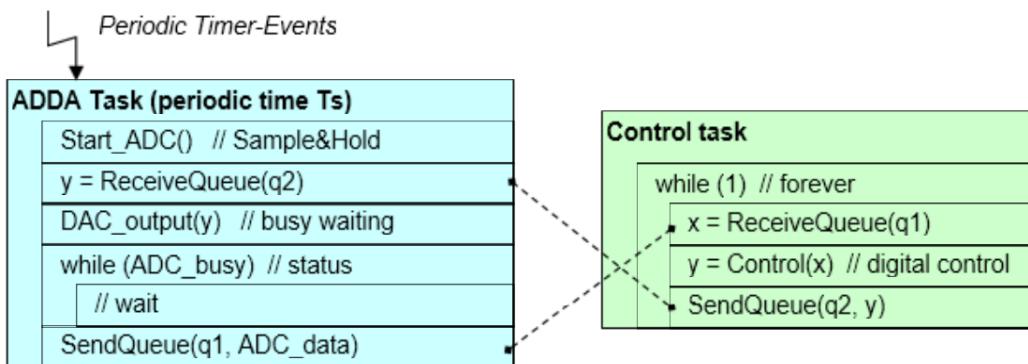
With digital sampling systems

1. strict periodicity of the sampling process in both the AD and the DA conversion
 2. to comply with the Nyquist condition (the analog lowpass filters - before any ADC and -after any DAC- with the right cutoff frequency)
- shall be respected.

Digital signal processing of analog signals involves very strict requirements to real-time conditions of some software tasks, due to the periodic and exact sampling conditions.

Action	Time Condition
input of a sample of an analog signal	
computation of a signal sample	
output of the calculated signal sample	

Example: Sampling Control (Digital Control) as real-time application using 2 Tasks and 2 Queues



*the Queue q2 must be initialized, before the ADDA Task starts.

Instead of a queue, often just a semaphore is used for synchronization

Negative Example: Not appropriated due to differences in run-time jitter !!

- → Non deterministic sampling
Violation of the exact real-time condition !!
- → Samples cannot be reconstructed,
Distortions and Artefacts with (HQ) Multimedia signals
- → Stability of a closed-loop control is in danger, cannot be guaranteed !!

```

while (1)
    Start_ADC() // Sample&Hold
    DAC_output(y) // busy waiting
    while (ADC_busy) // status
        // wait for ADC_data
    y = Control(ADC_data) // digital control
    delay(Ts-Tx) // constant delay

```

Not like that !!

Quantization

A discrete-time signal $x(k \cdot T_s)$ is quantized using an AD converter, i.e. its value (amplitude) is approximated by an integer number with N discrete values. This is because a digital computer can handle value discrete (integer) values only, not value-continuous (real) values.

The number of values N are often powers of two, i.e. $N = 2^M$, so one obtains an M -bit AD conversion, with the conversion factor c_x , which gives the value of an LSB (least significant bit) as a physical quantity

$$x_Q = Qx \cdot c_x + x_{\min} = Qx \cdot \frac{x_{\max} - x_{\min}}{N} + x_{\min}$$

$$c_x = \frac{x_{\max} - x_{\min}}{N} = \frac{x_{\max} - x_{\min}}{2^M}$$
(3.1)

x_Q integer representation of x

x_Q quantized physical quantity

c_x conversion factor

In automation 8-, 10-, 12-, 14- bit AD converters are used more often than, say, 16-, 20- or 24-bit ADCs, the latter are mainly used for audio signals.

The deviation $e = x_Q - x$ is called quantization error. In most applications, the number of conversion stages is N are chosen high enough, such that the quantization error is negligible, then $x_Q = x$.

Example: A pressure sensor with an 8-bit AD converter is equipped for a measuring a pressure signal with a range of values $(p_{\min}, p_{\max}) = (0, 1500)$ mbar.

The conversion factor by (??) is $c_p = 1500 \text{ mbar} / 2^8 = 5.86 \text{ mbar}$.

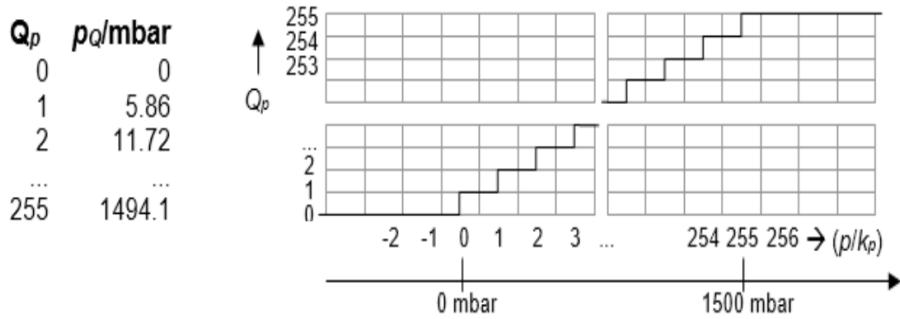
The quantization of the sensor signal is a step function with a linear increase. The value change takes place usually with a so-called **mid tread** pattern, i.e., in the middle of an interval.

The pressure signal is to be scanned by a process computer at a sampling frequency of 10 Hz. A warning signal at a low pressure $p_L = 0.7$ bar and at overpressure $p_H = 1.2$ bar shall be issued.

The bandwidth of the analog low-pass filter must be less than 5 Hz!

An appropriate real-time program for a microcontroller shall be designed with two tasks: 1. a task for pressure signal acquisition, 2. analysis and output.

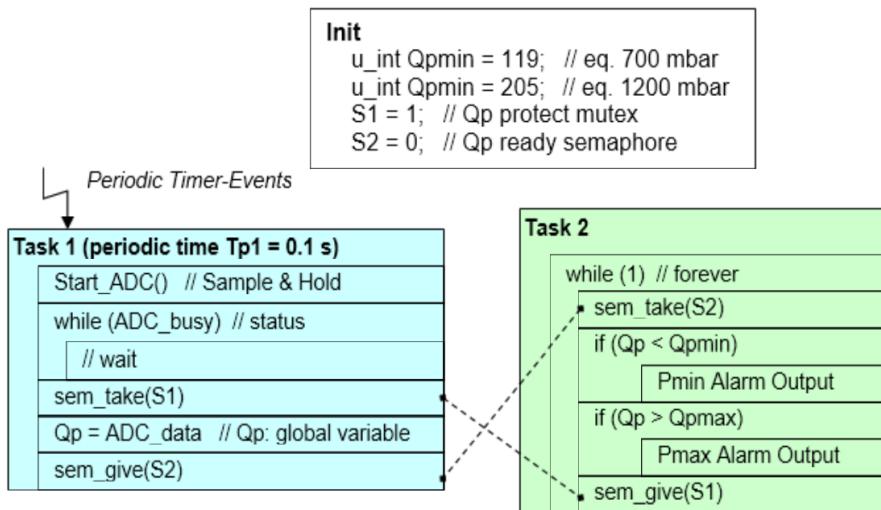
Possible Solution: 2 Tasks in an FPN schedule:



- **task 1:** $T_{p1} = 100$ ms, (sampling jitter < 0.1 ms). Acquisition of values Qp , setting the semaphore S1, if a new value $Qp[n]$ is available.
- **task 2:** wait for S1, then

(a) evaluate $Qp[n]$: compare with the lower limit
 $Qp_{min} = \text{round}(p_{min} / k_p) = \text{round}(700 \text{ mbar} / 5.86 \text{ mbar}) = 119$
set Output_pmin, if ($x[n] < Qp_{min}$)

(b) evaluate $Qp[n]$: compare with the upper limit
 $Qp_{max} = \text{round}(p_{max} / k_p) = \text{round}(1200 \text{ mbar} / 5.86 \text{ mbar}) = 205$
set Output_pmax, if ($x[n] > Qp_{max}$)



3.1.3 Digital Filters

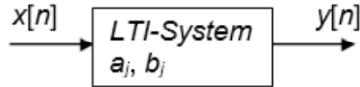
Linear digital filters can be described as LTI-Systems (linear, **time-invariant**) by means of difference equations. The general form of a linear difference equation with constant, real coefficients a_j, b_j is

$$a_0y[n] + a_1y[n-1] + a_2y[n-2] + \dots + a_Ny[n-N] = b_0x[n] + b_1x[n-1] + b_2x[n-2] + \dots + x_Mu[n-M]$$

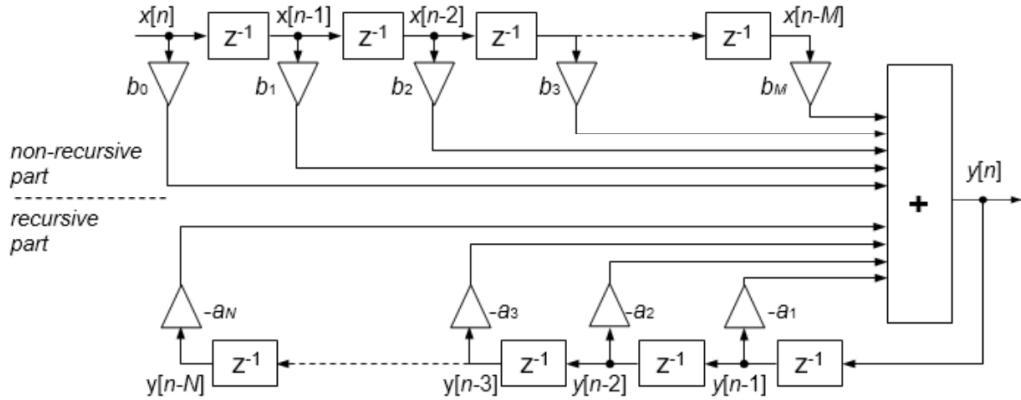
$$\sum_{j=0}^N a_j \cdot y[n-j] = \sum_{j=0}^M b_j \cdot x[n-j]$$

it is always possible to define $a_0 = 1$, which results in a recursion formula for the actual output value with constant coefficients a_j, b_j

$$y[n] = \sum_{j=0}^M b_j \cdot x[n-j] - \sum_{j=1}^N a_j \cdot y[n-j] \text{ with } a_0 = 1 \text{ (by convention)} \quad (3.2)$$



Structure of the general LTI-System



The z^{-1} rectangles symbolize a delay by one sample in time, the triangles symbolize a multiplication with a constant. The denominator degree N is also known as filter order. There are 2 basic types of digital filters

1. recursive filters with **IIR**-behavior (infinite impulse response), with $N > 0$
2. non-recursive filters with **FIR**-behavior (finite impulse response), with $N = 0$

With each new sampling period (increment of index k) a new output value $y[n]$ can be started earliest after the availability of a new input value $x[n]$, and the calculation needs to be finished before the next sampling period (Deadline !) see 3.1.1

For Integer implementations it is often economical to scale the difference equation (??) with an appropriate factor S such, that the multiply and add operations can be realized with integer arithmetic operations, most microcontroller CPUs support (rather than floating point arithmetic, which is common to larger, powerful CPUs):

$$y[n] = \frac{\sum_{j=0}^M (S b_j) \cdot x[n-j] - \sum_{j=1}^N (S a_j) \cdot y[n-j]}{S} = \frac{A}{S}, \quad S \in \mathbb{N} \quad (3.3)$$

The coefficients Sb_j and Sa_j can be represented with sufficient precision by integer numbers, if S is chosen large enough. Since input- and output sequence values $x[]$ and $y[]$ are also integers, the numerator in (??) can be evaluated using integer-multiplication and -addition using an **accumulator** A with increased width (i.e. len x,y: 16-bits, len A: 64 bit).

Particularly effective as a choice for $S = 2^{ldS}$ is a power of 2, thus the final division by N can be replaced by an equivalent right-shift operation

$$\frac{A}{S} (A >> ldS) \quad \text{without round op.}$$

$$\frac{A + S/2}{S} (A + S/2) \gg ldS \quad \text{with round op.}$$

Rules for Optimizing Execution Time on Standard-Microcontrollers

Since most standard micro-controller CPUs do not support floating point arithmetic in hardware, and also lack for a full division (modulo) hardware support, these functions are implemented by means of software libraries. With critical CPU load, a code optimization is useful to reduce the WCET of any computationally intensive task. Particularly effective measures are

1. Avoiding add/sub and multiplication (float, double)
→ scaling by a power of two, and multiply with integers
Exception: floating point operations are fine, if supported by the hardware.
2. Avoiding of Division (/)
→ Right Shift ($>>$) with powers of two instead of division
3. Avoiding of Modulo (%)
→ Bitwise AND (&) with powers of two instead of modulo
4. Review of the assembly based on the compiler list files (based on the specified CPU cycles) for execution time critical code

Example Digital IIR-Lowpass 1. Order

$$y[n] = b_0 \cdot x[n] - a_1 \cdot y[n-1] \quad y[n] = b_0 \cdot x[n] - a_1 \cdot y[n-1]$$

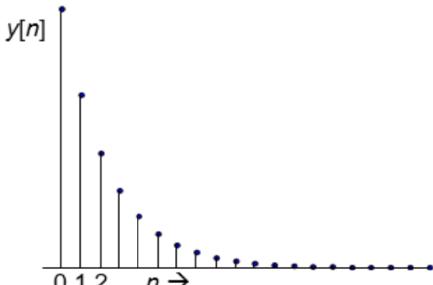
Given: filter-coefficients: $b_0 = 0.3333$, $a_1 = -0.6667$.

Wanted: impulse-response and step-response for $n \geq 0$. C-realization with float- and integer- (fixed-point) arithmetic.

- Impulse response $x[n] = [n]: x = 1, 0, 0, 0, \dots$

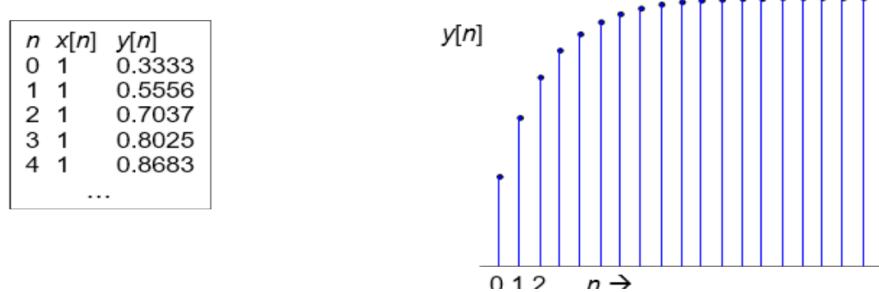
n	$x[n]$	$y[n]$
0	1	0.3333
1	0	0.2222
2	0	0.1481
3	0	0.0988
4	0	0.0658
...		

→ Impulse-response is infinite long → IIR



Impulse-response is infinite long → IIR

- Step-response $x[n] = [n]: x = 1, 1, 1, 1, \dots$



→ The step-response approximates the value 1.0 for large n , as expected.

C- Realization (float):

On microcontrollers without float-hardware needs hundreds of machine cycles and is thus very un-effective (section 3.2.1) !

Filter(x)	// IIR Filter: $y[n] = b_0 \cdot u[n] - a_1 \cdot y[n-1]$ // ----- short IIR1 float(short x) // floating point realization
b0 = 0.3333; // non-recursive	const float b0 = 0.3333;
a1 = -0.6667; // recursive coeff	const float a1 = -0.6667;
y = ADC_in * b0	float y = 0;
y -= filter_output * a1	extern short filter_output; // global variable
filter_output = y // global	y += x * b0; y -= filter_output * a1; filter_output = y; return y; // floating point: very unefficient !

C- Realization (integer):

Using the integer difference equation (??) with a power of 2: $S = 2^{ldS}$

$$\begin{aligned} S \cdot y[n] &= (S b_0) \cdot x[n] - (S a_1) \cdot y[n-1] \\ S b_0 &= S \cdot 0.3333 \\ -S a_1 &= S \cdot 0.6667 \end{aligned}$$

For example, if $ldS = 12$ is chosen, $S = 2^{12} = 4096$ follows

$$\begin{aligned} 4096 \cdot y[n] &= 1365 \cdot x[n] - (-2761) \cdot y[n-1] \\ S b_0 &= 1365 \\ S a_1 &= -2761 \\ y[n] &= \frac{1365 \cdot x[n] + 2761 \cdot y[n-1]}{4096} = \frac{A}{4096} \approx (A >> 12) \end{aligned}$$

Thus, with the scaled difference equation the S -fold result is calculated and stored in the accumulator A. Instead of dividing A by S, the effective right-shift operation (here: $>> 12$) is used.

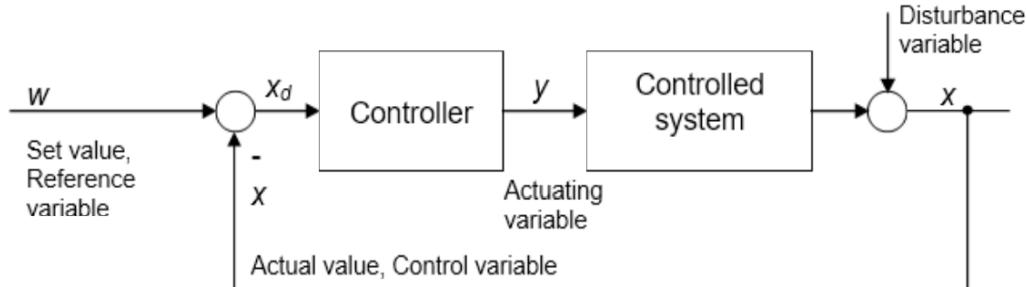
Filter(x) <pre>b0 = 0.3333; // non-recursive a1 = -0.6667; // recursive coeff y = ADC_in * b0 y -= filter_output * a1 filter_output = y // global</pre>	<pre>// ----- // IIR Filter: y[n] = b0*u[n] - a1*y[n-1] #define ldS 12 #define S (1 << ldS) // Skaling factor short IIR1 int(short x) // (fixed point) realization { const int Sb0 = 0.3333*S; // = 1365 const int Sa1 = -0.6667*S; // = -2731 int A; extern short filter_output; // global variable A = x * Sb0; // int promotion A -= filter_output * Sa1; // int promotion filter_output = (A >> ldS); // avoid div ! return filter_output; }</pre>
--	---

Even on microcontrollers without float-hardware, but hardware multiplier for integers utilizes only a handful of machine cycles and thus is very effective (on AVR ca. factor 100 faster than the float-version) !

3.1.4 Digital Controls

PID Controller

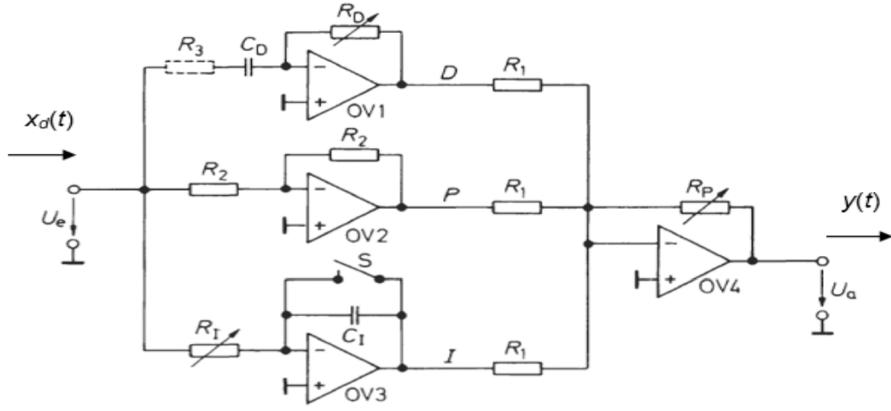
For many control systems in engineering, controls are used, which follow the so-called PID (**P**roportional-**I**ntegral-**D**ifferential) principle. Each of the 3 controller parts has its term with a specific parameter k_R , T_N , and T_V which allow the system response to be optimized with respect to stability, or under the influence of disturbances



System behavior of an analog PID Controller

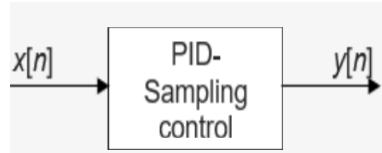
$$y(t) = K_R \cdot \left[x_d(t) + \frac{1}{T_N} \int x_d(t) dt + T_V \frac{dx_d(t)}{dt} \right] \quad (3.4)$$

For an analog PID controller the circuit principle below was chosen in the past, the parameters were often calculated approximately. Finally, with the complete system, the parameters were finely adjusted empirically by adjustment of the balancing resistors.



dimensioning: $k_R = \frac{R_p}{R_1}$, $T_N = R_I \cdot C_I$, $T_V = R_D \cdot C_D$ [Tietze]

The PID controller behavior (??) can be simulated using a digital sampling system with a suitable control algorithm invoked periodically with the sampling period T_s . The derivation is shown in [Oppen] oder [Kamm]:



Standard PID Regler Algorithmus ([LutzW] p. 477)

$$y[n] = y[n - 1] + K_R \cdot \left[x[n] \cdot \left(1 + \frac{T_V}{T_s} \right) - x[n - 1] \cdot \left(1 - \frac{T_s}{T_N} + 2 \frac{T_V}{T_s} \right) + x[n - 2] \cdot \frac{T_V}{T_s} \right] \quad (3.5)$$

PI-Controller

If the time constant T_V of the differential term is set to 0 in (??), the algorithm for the PI-controller is obtained, which has a proportional- and an integral term only:

$$y[n] = y[n - 1] + K_R \cdot \left[x_d[n] - x_d[n - 1] \cdot \left(1 - \frac{T_s}{T_N} \right) \right] \quad (3.6)$$

The PI controller type is quite popular, it can be realized with the PID-control algorithm.

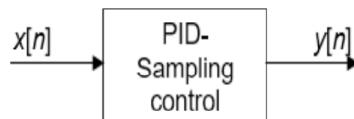
P-Controller

A P-controller has a proportional term only, with the simple algorithm

$$y[n] = K_R \cdot x_d[n] \quad (3.7)$$

PID-Controller as Digital Filter

The PID-control algorithm (3.5) can be expressed as a digital filter in



$$[y[n] = \sum_{i=1}^M a_i \cdot y[n-i] + \sum_{k=0}^N b_k \cdot x[n-k]] \quad (3.8)$$

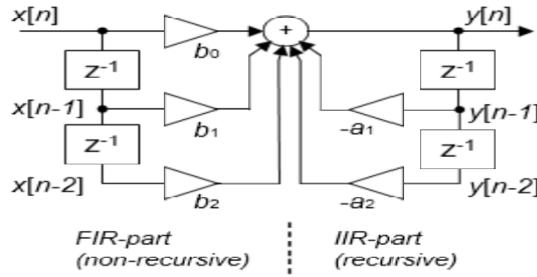
a common direct form (here, with input sequence $x_d[n] = x[n]$)

The coefficients for the des PID algorithm (??) thus define the coefficients of an IIR (Infinite Impulse Response) filter of 2. order

$$a_1 = -1, a_2 = 0, b_0 = k_R \cdot \left(1 + \frac{T_V}{T_S}\right), b_1 = -k_R \cdot \left(1 - \frac{T_s}{T_N} + 2\frac{T_V}{T_S}\right), b_2 = k_R \cdot \frac{T_V}{T_S}$$

$$y[n] = \sum_{k=0}^2 b_k \cdot x[n-k] - \sum_{i=1}^1 a_i \cdot y[n-i] = b_0 \cdot x[n] + b_1 \cdot x[n-1] + b_2 \cdot x[n-2] - a_1 \cdot y[n-1]$$

A digital filter on 2. order in direct form can be represented as signal flow for time discrete signals (sequences), with triangles for multiplication with coefficients, the z^{-1} rectangles the delay elements for 1 sample and the circles represent additions.



The algorithm (3.8) calculates a new output value $y[n]$, for each new sampling period T_S with a new input value $x[n]$ and two previous input values $x[n-1]$ and $x[n-2]$ and the last output value $y[n-1]$, which is stored.

For calculating digital filters on microcontrollers, it is generally advantageous to use optimized libraries (if available), which respect the specifics of the processor CPU. With optimizing digital filter algorithms significant reductions in execution time can be achieved (see 3.2.1).

PID Algorithm in C

The algorithm (3.8) can be programmed in its general form in floating-point arithmetic easily with the C programming language:

- **IIR2()** realizes (3.8), which is the actual digital filter, and, buffering of the 2 past output values $y[n-1]$ and $y[n-2]$
- **store_xd()** realizes the control difference $x_d[n]$ from the current process value $x[n]$ and setpoint $w[n]$. **store_xd()** also realizes the buffering of the 3 controller input values $x_d[n], x_d[n-1], x_d[n-2]$
- **init_pid()** calculates from a given set of controller parameters k_R, T_N , und T_V with the constant sampling period T_S a set of filter coefficients $a_pid[]$ und $b_pid[]$ by (3.8)

```

1 // -----
2 float a_pid[3], b_pid[3], xd[3] = {0., 0., 0.}, y[3] = {0., 0., 0.}, w = 0.;
3 // -----
4 void init_pid(float a[], float b[], float kr, float Tn_Ts, float Tv_Ts)
5 {
6     a[0] = a[2] = 0.0;    // Tn_Ts = Tn / Ts
7     a[1] = 1.0;          // Tv_Ts = Tv / Ts
8     b[0] = kr*(1.0 + Tv_Ts);
9     b[1] = -kr*(1.0 - 1.0/Tn_Ts + 2.0*Tv_Ts);
10    b[2] = kr*Tv_Ts;
11 }
12
13 // -----
14 void store_xd(float x0, float w)
15 {
16     xd[2] = xd[1];      // xd*z^(-2)
17     xd[1] = xd[0];      // xd*z^(-1)
18     xd[0] = w-x0;       // xd*z^(0)
19 }
20
21 // -----
22 float IIR2(float x[], float y[], float a[], float b[])
23 { // --- IIR 2nd order x[n] -> (a,b) -> y[n]
24     float sum;
25
26     // non-recursive (FIR) section
27     sum = x[0]*b[0] + x[1]*b[1] + x[2]*b[2];
28
29     // shift output buffer values
30     y[2] = y[1]; // y*z^(-2)
31     y[1] = y[0]; // y*z^(-1)
32
33     // recursive (IIR) section
34     sum += y[1]*a[1] + y[2]*a[2];
35
36     return y[0] = sum; // y*z^(0)
37 }

```

It should be noted, that the use of floating-point arithmetic in microcontrollers without hardware floating-point unit usually results in large execution times, and is therefore usually avoided entirely → use **integer arithmetic** (3.2.1) with less powerful microcontrollers !

As real-time software realization, it can be favorable, to calculate the FIR and the IIR filter parts by concurrent tasks, which can then be run simultaneously on a multiprocessor hardware.

Requirements for the Digital PID-Controller

Since this controller is a digital sampling system, it must respect the requirements of the sampling theorem on the strict periodicity with T_s with sampling of the control variable x and the output of the actuating variable y .

For the input signal, an AD converter, either working as SAR (usually integrated on microcontrollers) or as flash type, a good strategy for getting an ADC value within the strict timing must be found (starting conversion, waiting on the result).

1. Events with period T_s and with **minimal deviation (jitter)** must be produced definition of a timer event (T_s)
2. start of AD conversion for the **input** must occur immediately after the T_s -event, **minimal jitter**, (limited by the specified interrupt latency).
3. The **output** value must be DA converted (with a delay of exactly one sampling period), with minimal jitter aas with AD conversion.
4. It is important to ensure that the algorithm operates only with a **consistent set** of parameters and input and output buffers with consistent values.
5. The **controller parameters** must be **adjustable** from outside (e.g. by asynchronous bus event). The latency time allowed for a complete switch of the parameters is $100 \cdot T_s$. It must be ensured, that any parameter change, always results with a stable digital filter.
6. The set value must be adjustable from outside (e.g. by asynchronous bus event). The latency time allowed here is $20 \cdot T_s$.
7. The real-time program shall be executable with any type of task scheduling.

Task Definition

A real-time program with 4 tasks can meet the specified requirements:

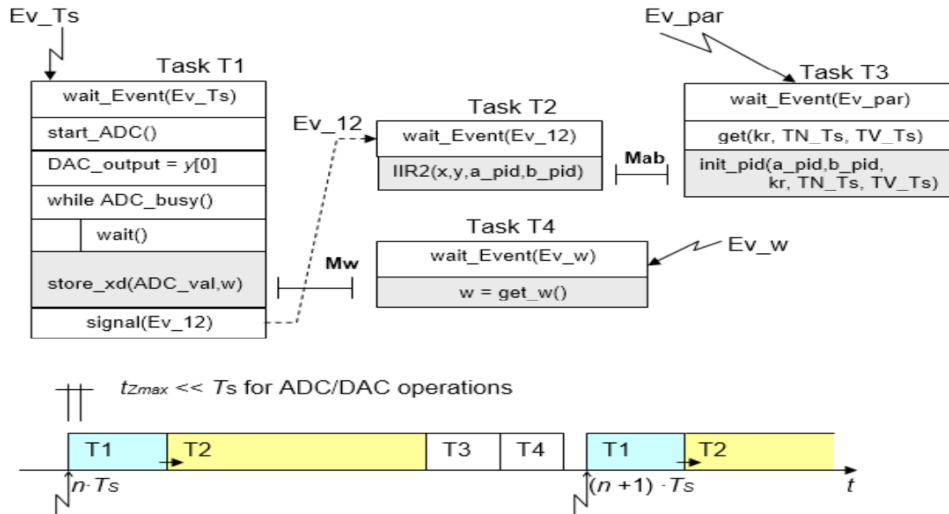
- **Task T1:** a high-priority task, which controls the timing conditions for reading the input from the AD converter and writing the output to the DA converter. (wait on periodic timer event **Ev_Ts**)
- **Task T2:** a medium priority task, which performs the calculation of the output sample by the IIR algorithm, after a new input value is ready. Task 2 may run only when task 1 is finished → event **Ev_12**.
- **Task T3:** an asynchronous low-priority task, which performs the conversion and modified PID control parameters and provides a consistent set of IIR filter coefficients → event **Ev_par**.
- **Task T4:** an asynchronous low-priority task, which lets change the set value → event **Ev_w**

Critical Data

- **w (set value):** read by T1 and can be concurrently written by T4 → Mutex **Mw**
- **a_pid, b_pid PID filter coefficients:** These are calculated and read by T2, concurrently written by T3 → Mutex **Mab**

Although the input buffer $xd[]$, and the buffer of output values $y[]$ is accessed by the two tasks T1 and T2, it is guaranteed by the order synchronization of T1-T2, that there will be no conflict, even without a mutex.

Realization of a Digital PID Control with 4 Tasks



This real-time program

- assures with T1: compliance with the strict time conditions for the two sampling systems ADC \rightarrow x[0] and y[0] \rightarrow DAC: called periodically with minimal jitter.
- by **Ev_T12** is ensured, that T2 runs only when T1 is finished (cooperation sync)
- by Mutex **Mab** is ensured, that the PID algorithm is executed only with a consistent set of parameters.
- assumption is, that the utilization of task 2 was determined by measuring WCET to be $H_2 \approx 50\%$. If H_2 is too large, optimizations are necessary (e.g. integer arithmetic)

Exercise 1: Write C Code for Realizing Task T2 under FreeRTOS.

- develop by re-factoring PingPong, realize $Ts = 2$ ms

```

1  /*
2   * -----
3   * iir2.c: Module for realization of a digital filter, 2. order
4   * -----
5   float a_pid[3], b_pid[3], xd[3] = {0., 0., 0.}, y[3] = {0., 0., 0.}, w = 0. ;
6   // -----
7   float IIR2(float x[], float y[], float a[], float b[])
8   { // --- IIR 2nd order x[n] -> (a,b) -> y[n]
9     float sum;
10    // non-recursive (FIR) section
11    sum = x[0]*b[0] + x[1]*b[1] + x[2]*b[2];
12    // shift output buffer values
13    y[2] = y[1]; // y*z^(-2)
14    y[1] = y[0]; // y*z^(-1)
15    // recursive (IIR) section
16    sum += y[1]*a[1] + y[2]*a[2];
17    return y[0] = sum; // y*z^(0)
18 }
19
20 // -----
21 // -----
22 #include "FreeRTOS.h"

```

```

23 xSemaphore sEv_12, siir;                                // Semaphore sEv_12, Mutex siir
24 // -----
25 void vT2(void * pvParameters)                         // re-factored vPingTask
26 {
27     while(1) {
28         xSemaphoreTake(sEv_12, portMAX_DELAY); // on Event Ev_12
29         xSemaphoreTake(siir, portMAX_DELAY);   // acquire mutex
30         IIR2(xd, y, a, b);                  // Filter operation
31         xSemaphoreGive(siir);               // release mutex
32     }
33 }
34
35 // -----
36 void vT1(void * pvParameters)                         // re-factored vPrintTask
37 {
38     short ADC_in;
39     portTickType xLastWakeTime;
40     xLastWakeTime = xTaskGetTickCount();
41
42     while(1) {
43         vTaskDelayUntil(&xLastWakeTime, 2*configTICK_RATE_HZ/1000); // exact
44         periodic Ts = 2 ms
45         ADC_start();           // start AD (--> SAR sample & hold) ...
46         ...                   // busy waiting stuff here (like DAC outputs)
47         while(ADC_busy());    // wait for ADC (some microseconds) ...
48         ADC_in = ADC_get();   // store ADC result
49         update_x_buffer(ADC_in); // update the filter input buffer xd[]
50         xSemaphoreGive(sEv_12); // sync to task T2
51     }
51 }

```

Exercise 2: PI controller (for ATmega microcontrollers)

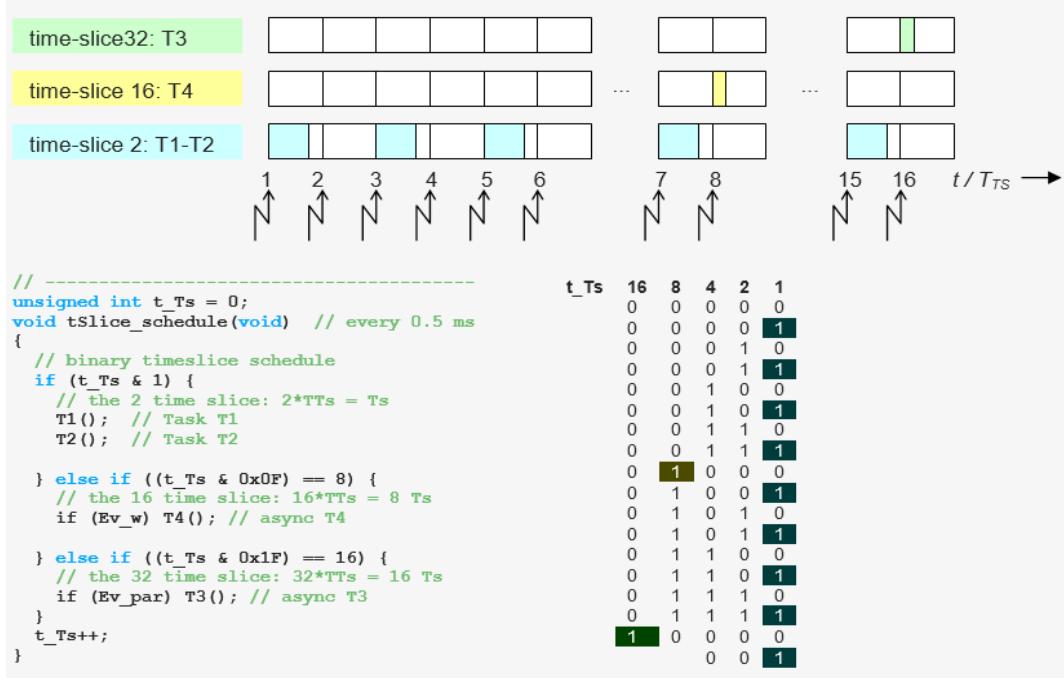
1. Design a real-time program for a PI controller with the non-preemptive binary fixed time-slice method (FTS) for the following parameters
 - execution times: $T_{e1} = 50 \mu\text{s}$, $T_{e2} = 300 \mu\text{s}$, $T_{e3} = 100 \mu\text{s}$, $T_{e4} = 50 \mu\text{s}$.
 - T3 und T4 shall be able to process asynchronous events about 20 times per sec.
 - the initial control parameters: $k_R = 5$, $T_N = 20 \text{ ms}$.
2. Determine a new Schedule for the case where due to halving of the clock frequency of the microcontroller, each task has approximately twice the execution time of (1).

Solution:

$$1. H_1 = 5\%, H_2 = 30\%, H_3 = 0.1/20 = 0.5\%, H_4 = 0.05/20 = 0.25\% \quad H = 30.75\%$$

with 3 tasks, non-preemptive binary time-slice scheduler with $T_{TS} \geq \max(T_{ei})$,
 → periodic timer events with basic time-slice $T_{TS} = 0.5 \text{ ms}$, so execution times of each of the 4 tasks are below the maximum possible execution time of the basic time-slice.

- T1 and T2 run successively (no sync needed) in the time-slice called every 2 times
- T4 runs every 8 ms (= time-slice 16)
- T3 runs every 16 ms (= time-slice 32), all synchronous to the periodic timer events
- the execution times of all 4 tasks are well below the maximum time $T_{TS} = 0.5 \text{ ms}$.



2. Doubled execution time (e.g. due to reduction of the CPU clock frequency)

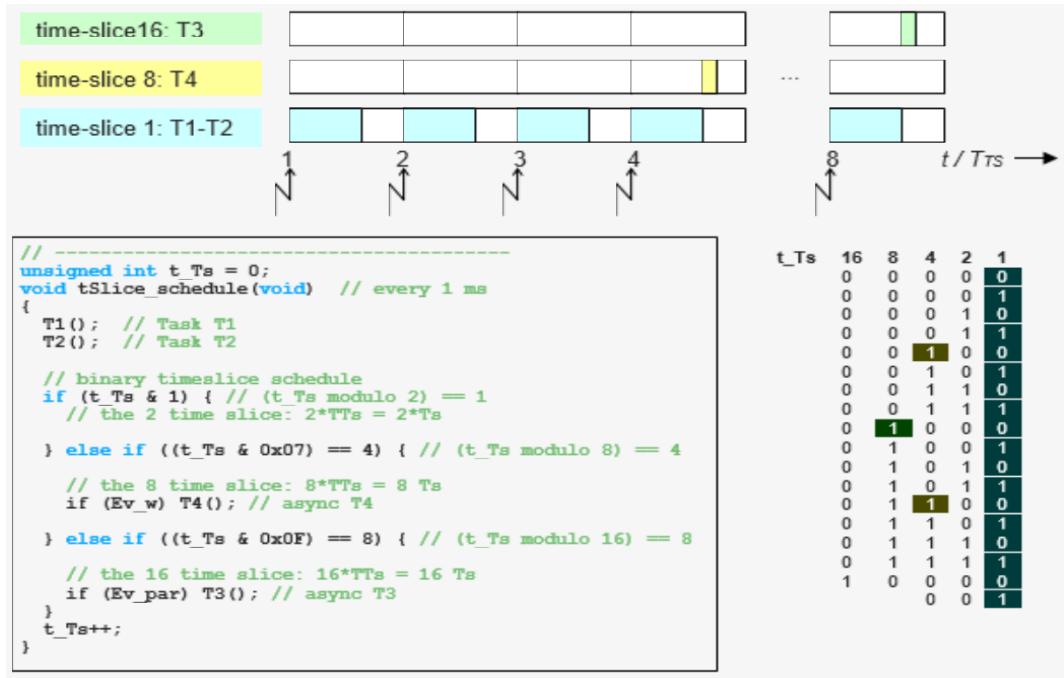
$T_{e1} = 100 \mu\text{s}$, $T_{e2} = 600 \mu\text{s}$, $T_{e3} = 200 \mu\text{s}$ und $T_{e4} = 100 \mu\text{s}$

$$H_1 = 10\%, H_2 = 60\%, H_3 = 0.2/20 = 1\%, H_4 = 0.1/20 = 0.5\% \quad H = 71.5\%$$

Solution:

with 3 tasks, non-preemptive binary time-slice scheduler with $T_{TS} \geq \max(T_{ei})$,

→ periodic timer events with basic time-slice $T_{TS} = 1.0 \text{ ms}$, so execution times of each of the 4 tasks are below the maximum possible execution time of the basic time-slice.



- T4 runs every 8 ms (=time-slice 8)
- T3 runs every 16 ms (=time-slice16), both asynchronous to the periodic timer events (after T1-T2) !
- The maximum execution time is within time-slice 8: $T_{e1} + T_{e2} + T_{e3} = 900 \mu\text{s}$.

3.1.5 Finite State Machines (FSM)

In most embedded applications finite state machines are used to interface with non-synchronous, periodic devices, such as switches, slow actuators, indication lamps, networks ...

Traditional event-based programming is used with modern GUI based standard OS, Windows, Linux, .. with both desktop and portable applications handling keyboard mouse, touchpanels, ... as inputs as well as with networks.→ https://en.wikipedia.org/wiki/Automata-based_programming

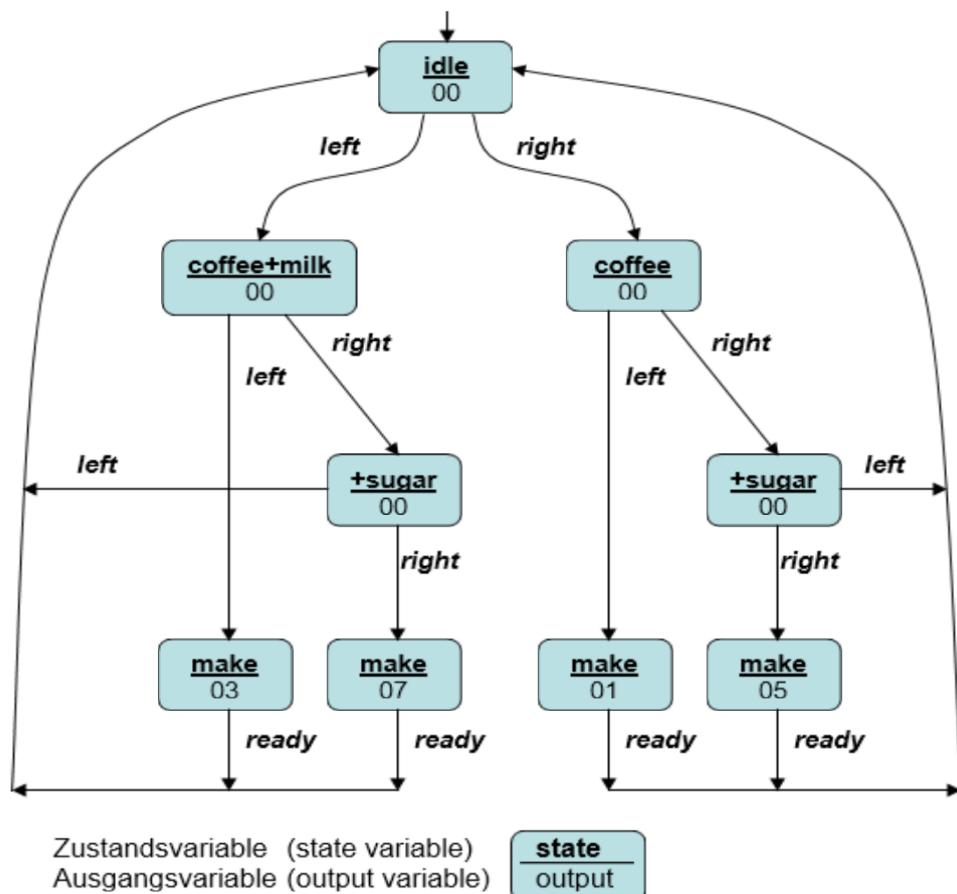


Figure 3.1.1: Finite State Machines (FSM)

Moore Finite State Machine: Each state has assigned exactly one output.

Coffee Automat: State Transient Table

state S	input	state S+	output
00 idle (initial)	left right	03: coffee+milk 01: coffee	0 0
01 coffee	left right	05: make coffee 03: coffee+sugar	0 0
02 coffee+milk	left right	06: make coffee+milk 04: coffee+milk+sugar	0 0
03 coffee+sugar	left right	00: idle 07: make coffee+sugar	0 0
04 coffee+milk+sugar	left right	00: idle 08: make coffee+milk+sugar	0 0
05 make coffee	!ready ready	05: make coffee 00: idle	01
06 make coffee+milk	!ready ready	06: make coffee+milk 00: idle	03
07 make coffee+sugar	!ready ready	07: make coffee+sugar 00: idle	05
08 make coffee+milk+sugar	!ready ready	08: make coffee+milk+sugar 00: idle	07

Coffee Automat: Implementation in C (→ ProgC FSM_CoffeeAutomat)

```

1 // -----
2 // the FSM states for the Coffee Automata
3 // -----
4 enum {  FSM_STATE_IDLE,
5         FSM_STATE_COFFEE,           FSM_STATE_COFFEE_MILK,
6         FSM_STATE_COFFEE_SUGAR,     FSM_STATE_COFFEE_MILK_SUGAR,
7         FSM_STATE_MAKE_COFFEE,      FSM_STATE_MAKE_COFFEE_MILK,
8         FSM_STATE_MAKE_COFFEE_SUGAR, FSM_STATE_MAKE_COFFEE_MILK_SUGAR,
9     };
10 // -----
11 int fsm_state = FSM_STATE_IDLE;
12 // -----
13
14 // -----
15 // the FSM outputs, one for each state, (Moore Automata model)
16 // -----
17 const int fsm_output[NUM_FSM_STATES] = { // output is fsm_output[fsm_state]
18     0,                                // bit0: add coffee
19     0, 0,                            // bit1: add milk
20     0, 0,                            // bit2: add sugar
21     1, 3,
22     5, 7
23 };
24
25
26 // -----
27 // the FSM inputs
28 // -----
29 enum {  FSM_INP_NONE=0, FSM_INP_LEFT,  FSM_INP_RIGHT,  FSM_INP_READY
30 // -----
31 int fsm_tic(int fsm_inp)    // called periodically, e.g. each 50 ms
32 {
33     // -----

```

```

34 // the FSM transients, realizing the state transient table
35 // -----
36 switch (fsm_state) {
37     case FSM_STATE_IDLE:
38         if (fsm_inp == FSM_INP_LEFT) fsm_state = FSM_STATE_COFFEE;
39         if (fsm_inp == FSM_INP_RIGHT) fsm_state = FSM_STATE_COFFEE_MILK;
40         break;
41
42     case FSM_STATE_COFFEE:
43         if (fsm_inp == FSM_INP_LEFT) fsm_state = FSM_STATE_MAKE_COFFEE;
44         if (fsm_inp == FSM_INP_RIGHT) fsm_state = FSM_STATE_COFFEE_SUGAR;
45         break;
46
47     case FSM_STATE_COFFEE_MILK:
48         if (fsm_inp == FSM_INP_LEFT) fsm_state = FSM_STATE_MAKE_COFFEE_MILK;
49         if (fsm_inp == FSM_INP_RIGHT) fsm_state = FSM_STATE_COFFEE_MILK_SUGAR;
50         break;
51
52     case FSM_STATE_COFFEE_SUGAR:
53         if (fsm_inp == FSM_INP_LEFT) fsm_state = FSM_STATE_IDLE;
54         if (fsm_inp == FSM_INP_RIGHT) fsm_state = FSM_STATE_MAKE_COFFEE_SUGAR;
55         break;
56
57     case FSM_STATE_COFFEE_MILK_SUGAR:
58         if (fsm_inp == FSM_INP_LEFT) fsm_state = FSM_STATE_IDLE;
59         if (fsm_inp == FSM_INP_RIGHT) fsm_state = FSM_STATE_MAKE_COFFEE_MILK_SUGAR;
60         break;
61
62     case FSM_STATE_MAKE_COFFEE:
63     case FSM_STATE_MAKE_COFFEE_MILK:
64     case FSM_STATE_MAKE_COFFEE_SUGAR:
65     case FSM_STATE_MAKE_COFFEE_MILK_SUGAR:
66         if (fsm_inp == FSM_INP_READY) fsm_state = FSM_STATE_IDLE;
67         break;
68
69     default:
70         ;    // no state change
71 }
72 OUTPUT(fsm_output[fsm_state]);      // Moore FSM: one output per state
73 return

```

The above implementation in C is optimized for readability rather than for minimum runtime or memory resources. However, the above code uses very little memory space and is running quite fast.

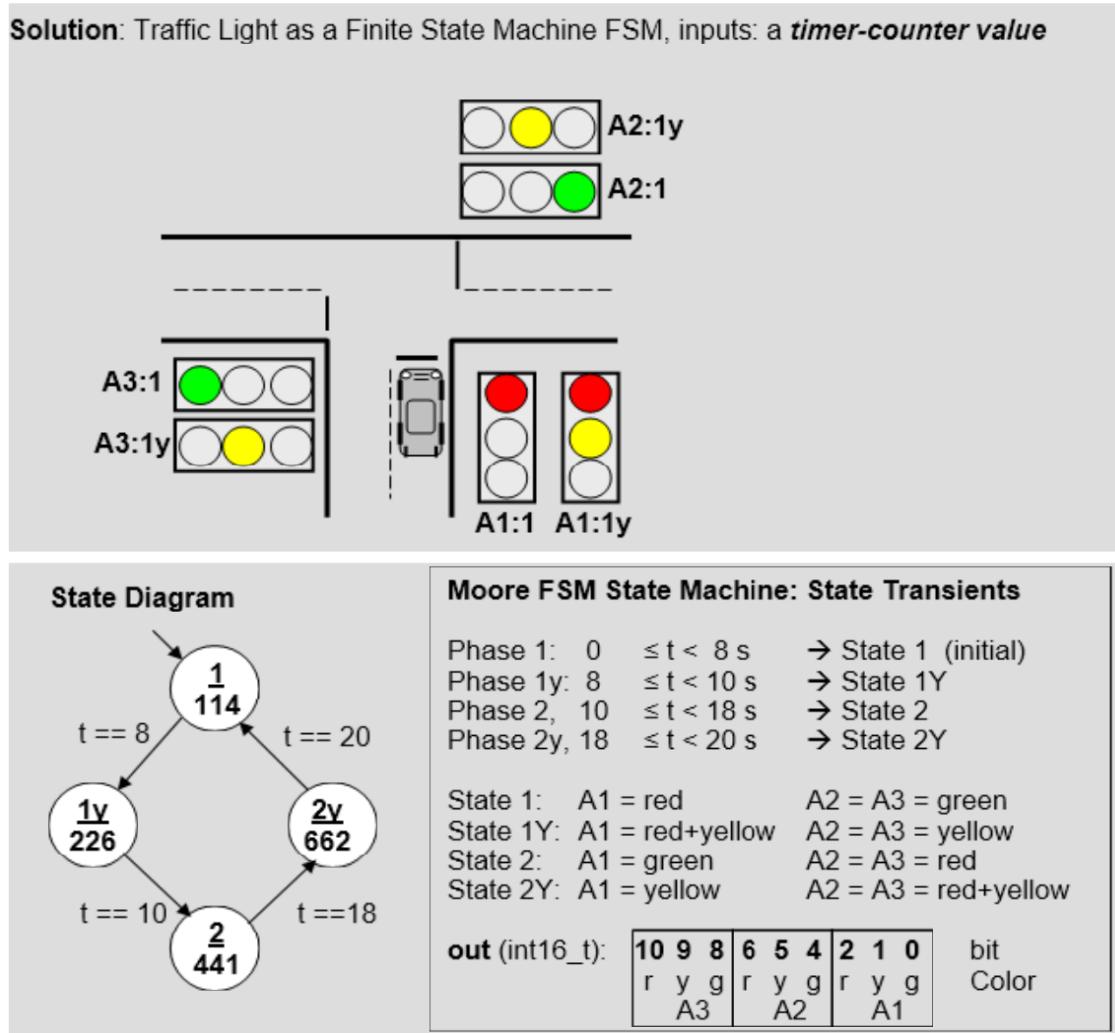
Recipe for Realizing an FSM in C (Moore FSM)

- Analyze the problem before coding, plot a state diagram, define the states
- Use enums to encode states in a well readable form

- Define outputs as enums with suitable encoding, e.g. as bit-fields
- Define input conditions causing the state transitions as enums as well
- Realize a periodically called fsm_tic function, realize transients and outputs

Exercise: Develop a Traffic-Light including yellow phases from section 1.2 as a Moore-FSM as the above Coffee Automat with C-Code. The yellow phases shall last 2 s, the main green/red phases shall be 8 s each. Assume the fsm_tic() is called with 1 s periodic time. Plot a State Diagram and the state transient table.

Solution: Traffic Light as a Finite State Machine FSM, inputs: a *timer-counter value*



C-Code (\rightarrow ProgC FSM_TrafficLight)

```

1 // -----
2 // the FSM states, here for the TrafficLight
3 // -----
4 enum fsm_state_e {
5     FSM_STATE_1=0,           FSM_STATE_1Y,
6     FSM_STATE_2,             FSM_STATE_2Y,
7     NUM_FSM_STATES
8 };
9
10 // -----
11 // the FSM state
12 // -----
13 int fsm_state = FSM_STATE_1; // initial state
14 int fsm_count = 0;          // the FSM counter
15 // -----
16 // the FSM outputs, bit assignments, one for each state (Moore)
17 // -----

```

```

18 enum fsm_out_e {  FSM_OUT_A1_G=1,    FSM_OUT_A1_Y=2,    FSM_OUT_A1_R=4,
19             FSM_OUT_A2_G=0x10,   FSM_OUT_A2_Y=0x20,   FSM_OUT_A2_R=0x40,
20             FSM_OUT_A3_G=0x100,  FSM_OUT_A3_Y=0x200,  FSM_OUT_A3_R=0x400,
21             NUM_FSM_OUTPUTS };
22
23 const int fsm_output[NUM_FSM_STATES] = {           // output is fsm_output[
24     fsm_state]
25     FSM_OUT_A1_R|FSM_OUT_A2_G|FSM_OUT_A3_G,      // output with FSM_STATE_1
26     FSM_OUT_A1_R|FSM_OUT_A1_Y|FSM_OUT_A2_Y|FSM_OUT_A3_Y,  // output with
27     FSM_STATE_1Y
28     FSM_OUT_A1_G|FSM_OUT_A2_R|FSM_OUT_A3_R,      // output with FSM_STATE_2
29     FSM_OUT_A2_Y|FSM_OUT_A2_R|FSM_OUT_A3_R
30         |FSM_OUT_A2_Y|FSM_OUT_A3_Y           // output with FSM_STATE_2Y
31 };
32 // -----
33 #define T_RED_GREEN 8 // Time for Red/Green phases in Ts
34 #define T_YELLOW 2   // Time for Yellow    phases in Ts
35 // -----
36 int fsm_tic()
37 {
38     // -----
39     // the FSM
40     // -----
41     if (fsm_count < T_RED_GREEN)
42         fsm_state = FSM_STATE_1;
43     else if (fsm_count < (T_RED_GREEN + T_YELLOW))
44         fsm_state = FSM_STATE_1Y;
45     else if (fsm_count < (2*T_RED_GREEN + T_YELLOW))
46         fsm_state = FSM_STATE_2;
47     else if (fsm_count < (2*T_RED_GREEN + 2*T_YELLOW))
48         fsm_state = FSM_STATE_2Y;
49     else {
50         fsm_state = FSM_STATE_1;
51         fsm_count = 0; // modulo cycle time
52     }
53     // -----
54     fsm_out();          // Moore FSM: one output per state
55     // -----
56     fsm_count++;
57     return fsm_state;
58 }
59 // -----
60 int main()
61 {
62     int ch;
63     gotoXY(0,0);
64     printf("Traffic light, Ctrl-Break: exit");
65     do { // basic terminal loop
66         ch = 0;
67         if (kbhit() ) {
68             ch = getch();
69         }
70         fsm_tic(); // fsm tick
71         Sleep(1000); // Ts = 1000 ms
72     } while ( ch != 127); // until user hits ctrl-backspace.
73     return

```

Chapter 4

Microprocessor Peripherals

4.1 Processor Core and Peripherals

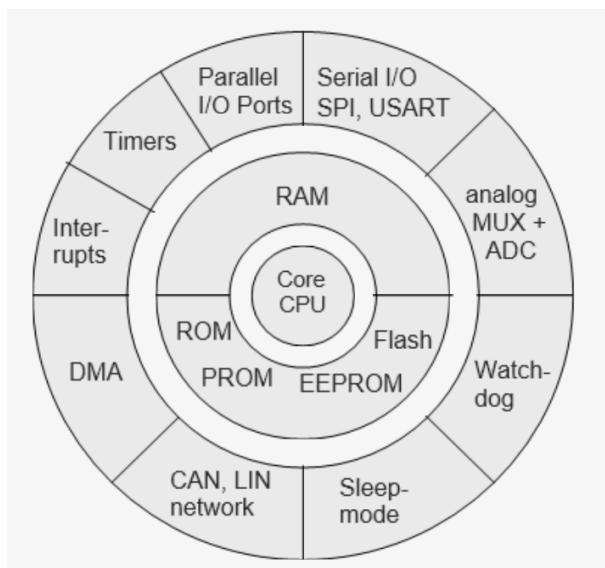


Figure 4.1.1: Layer model of a microcontroller (single chip) or a Microprocessor (more chips)

The microprocessor cores are often common CPU types used in the past as microprocessors. e.g. formerly: Motorola 680x0, 32 bit microprocessor (+ multi-board bodies) Today: Freescale "Coldfire" microcontroller = 68k CPU + integrated peripherals.

The single-chip solutions of modern microcontrollers are very cost effective and efficient, with already well-proven software → single chip solutions become more and more popular !

Microcontrollers are frequently used mobile, therefore energy management is of great importance (sleep mode).

With higher integration densities, microcontrollers are found today even with floating point or DSP - features (like integrated Multiply Accumulate hardware) for fast computation of digital filters, see section 3.3.4

Microcontrollers, as typically used in industrial applications have complex peripheral units like communication controllers for the CAN or LIN network.

The physical layer (ISO layer 1) of these networks though, is still implemented with external driver chips which, in high-voltage-technology (typ. 40V), while micro-controller chips are usually implemented in 5 V or 3.3 V technologies.

Example: AVR microcontroller AT90CAN

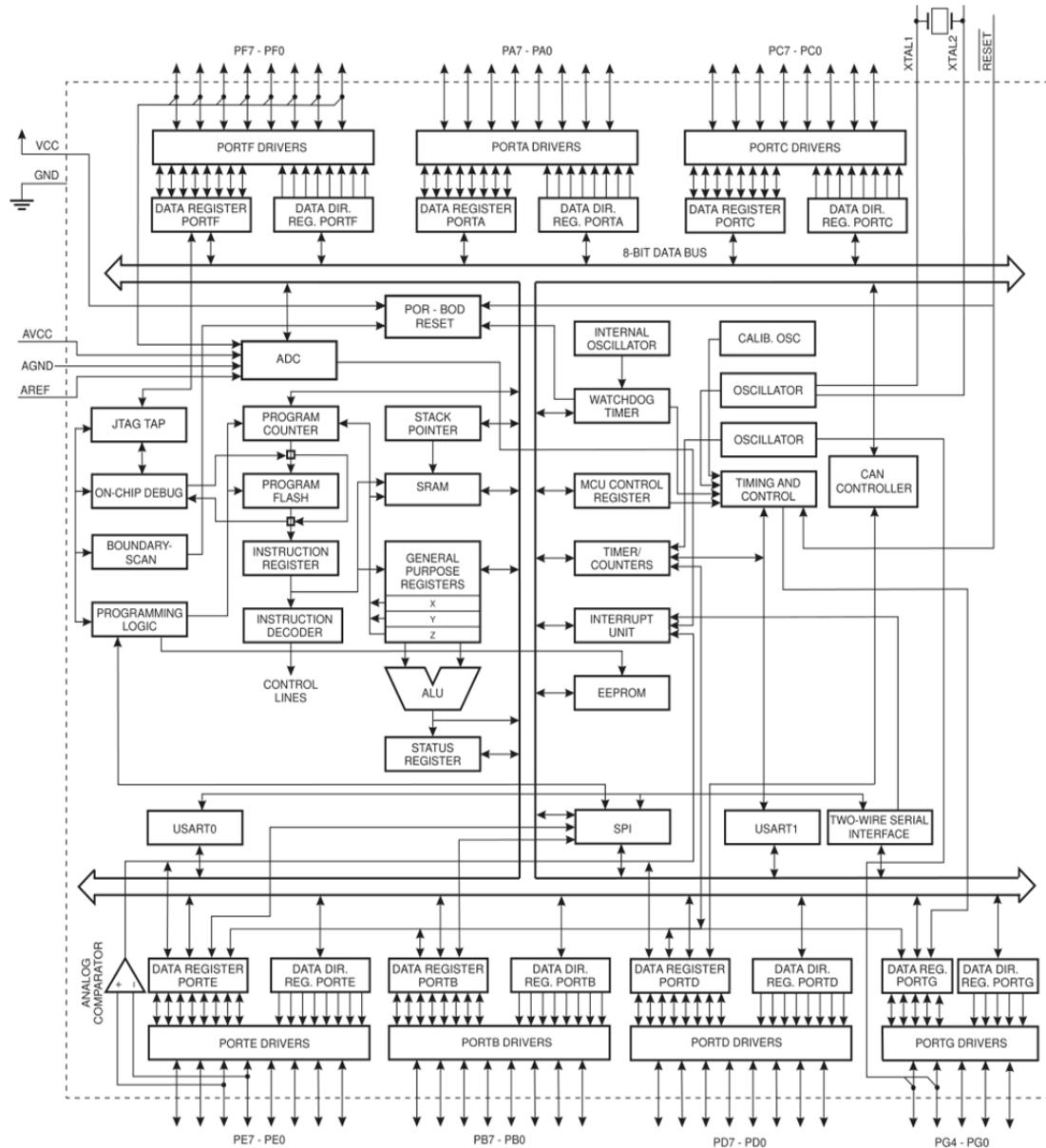


Figure 4.1.2: Layer model of a microcontroller (single chip) or a Microprocessor (more chips)

price \approx 1-2 € (2009, in larger volumes).

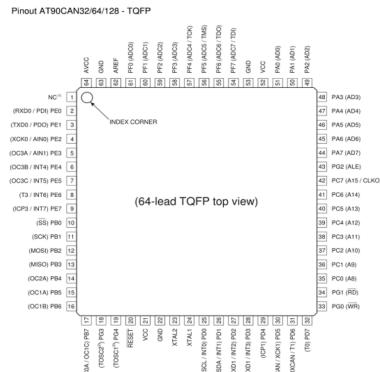


Figure 4.1.3: TQFP plastic housing 9 x 9 mm

4.1.1 Timers and Counters

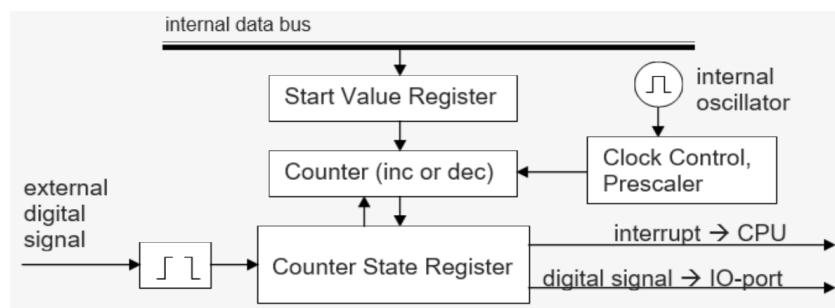


Figure 4.1.4: Basic structure of a timer-counter unit (CPU independent hardware)

Timers and counters are very important components for real-time operation: Fed from an internal (RC) or external (quartz, ceramic resonator) oscillator, interrupts can be triggered with minimal delay, if

- internal events are true, like equality to a predetermined counter value
- external events, like a rising or falling slope of a digital signal applied to some port can be stored as the count for this event in a register, and measuring at a later time as the time (counts) between such events.

Other important applications of the timer are PWM (Pulse Width Modulation) timers whose period length is usually fixed and the pulse width in terms of a short-term average can be controlled by a control register.

The short-term average value is changed, and an analog signal can be measured after an analog low pass filter. This filtering effect can be done for example by the inductance (winding) an electric motor, which filters the square wave voltage to a DC to a short time average with the same short-term arithmetic mean (see 4.4.2).

Example: Periodic Event with AVR TCNT0 Hardware Timer ISR(SIG_OVERFLOW0) (similar to the "AVR Timer Interrupts" exercise in the RTP-Lab)

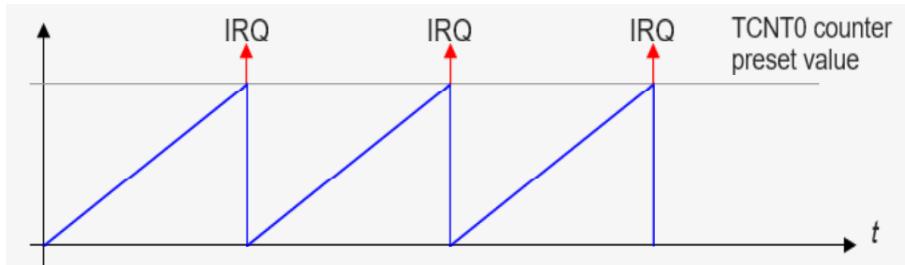
```

1 #include <avr\io.h>      // AVR Register and constant definitions
2 ...                      // (.. like #define TCNT0 0x32)
3 static unsigned int cnt; // the timeslice counter
4

```

```

5 #define TCNT_PRELOAD 62 // for f_timeslice = 2 kHz
6
7 void init() {
8     ...
9     CLKPR = 8; // prescaler Value
10    TCCR0B |= (1 << CS01); // Timer clock = system clock / CS
11        // CS = [1 8 256] mit [CS0 CS1 CS2]
12    TCNT0 = TCNT_PRELOAD; // Timer 0 Preload Value
13    ...
14    cnt = 0; // init timeslice counter
15 }
16
17 ISR(SIG_OVERFLOW0) // this function is called at timer 0 overrun
18 { // f_timeslice = fosc/CS(prescale)/TCNT_PRELOAD
19     TCNT0 = TCNT_PRELOAD; // next cycle Timer Preload Value (normal mode)
20     // time slice scheduling: 1/TTs = f_timeslice
21     if (cnt & 1) {
22         // the 2 time slice: 2*TTs (1 ms)
23         fastTask();
24     } else if ((cnt & 0x03) == 2) {
25         // the 4 time slice: 4*TTs (2 ms)
26         slowTask();
27     }
28     cnt++;
29 }
30
31 int main(void) // is executed after a reset
32 {
33     init();
34     do {
35     } while (1); // endless mainloop
36 }
```



4.1.2 Watchdog

The term watchdog is used for a component of a system, which monitors the function of other components. If a possible malfunction is detected, this can be either

- a signal according to system specifications (e.g. air bag warning light), or,
- appropriate steps are taken to fix the problem (e.g. reset).

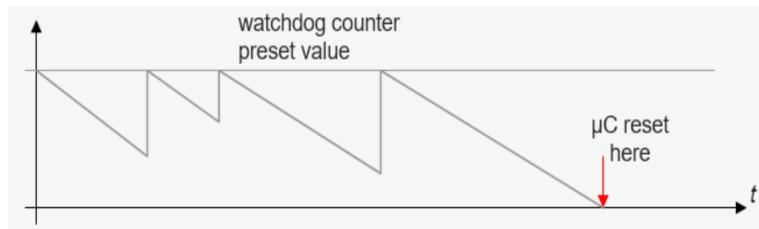
Specifically watchdogs in microcontroller-controlled electronic devices are used to avoid a complete failure of the device through *software failure*, or to take mitigating or workarounds. In most cases these are so-called time-out watchdogs, similar to a *dead-man's switch* with a locomotive.

Software-Watchdog

The software watchdog is an examining task running on the microcontroller. The watchdog checks, if all important tasks are executed correctly within in a given time frame, or, if a task needs an inadmissible long process time. This functionality can be realized by setting timeouts for individual tasks to avoid locks, if supported by the RTOS.

Hardware-Watchdog

Prevents the failure of an microcomputer system by triggering a special watchdog timer (hardware) by the software (one or more tasks) triggers at regular time intervals, to announce that the software still works properly (see AT90mega). With missing triggers, the watchdog timer resets the microcontroller (μC).



To achieve even higher security against errors in the μC , the same functionality as the internal watchdog timer can be realized by an external watchdog module, which is independent of the μC supply and its clock.

The watchdog module is a hardware counter with a separate integrated clock generation. The counter is set at regular intervals by the software through an I/O port to a predetermined preset value.

The counter is decremented in watchdog module, if it ever reaches 0, it may have come to a deadlock or some other critical situation within the software, which was not debugged before (**software failure (?)**).

However, the watchdog will reset the μC device to its defined initial state (as with power-on), such that the software system can work properly again.

Intelligent Watchdog (with increased security requirements)

With an intelligent watchdog, the microcontroller software has to answer a challenging message from an external watchdog correctly in an expected way challenge-response approach

In case of failure not only a reset of the microcontroller is generated, but the watchdog usually brings the whole system into a safe condition -regardless of the apparently uncontrolled microcontroller.

Example: Intelligent watchdog for the electronic accelerator (e-gas) in the car:

By careful choice of the challenges caused by arithmetic operations, the CPU's internal modules (ALU, program memory, working memory and registers are monitored).

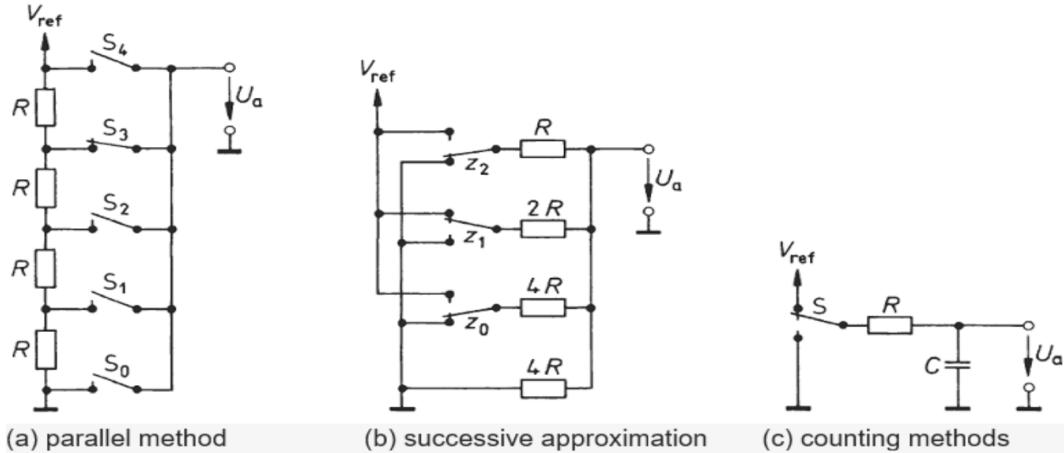
4.1.3 DA Converters

A voltage DA converter (DAC) converts an integer number Z (in a digital computer), into a voltage. The voltage U_a is proportional to the number Z (with a linear DAC), by

$$U_a = Z \bullet U_{LSB}$$

U_{LSB} is the voltage for the least significant bit (LSB), thus the voltage according to $Z = 1$.

One can distinguish three fundamentally different methods:



parallel (direct) method

with a voltage divider all possible output voltages are provided. With a 1-of-n decoder then closes the right switch, the desired output voltage is associated with.

successive approximation method (SAR)

each bit is associated with a switch. Using appropriately weighted resistors (often R-2R) the output voltage is setup by a loaded DC voltage divider.

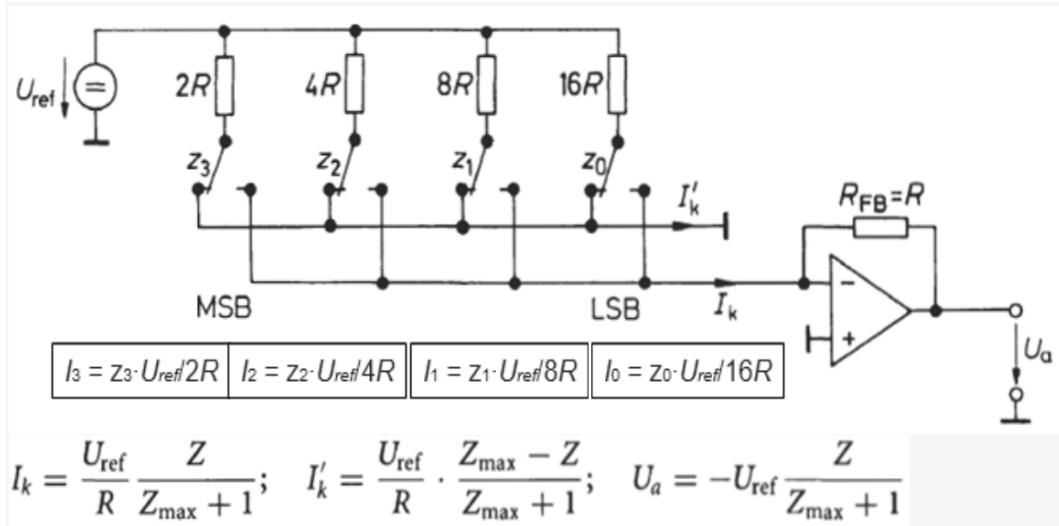
counting methods (PWM)

requires only a single switch, which is periodically opened and closed. Its duty cycle is controlled by a pulse width modulator such, that the arithmetic mean of the output voltage corresponds to the desired value.

The comparison of the three methods shows that the parallel method (a) requires Z_{max} switches, the SAR type needs $\log(Z_{max})$ switches, and the PWM type only one switch.

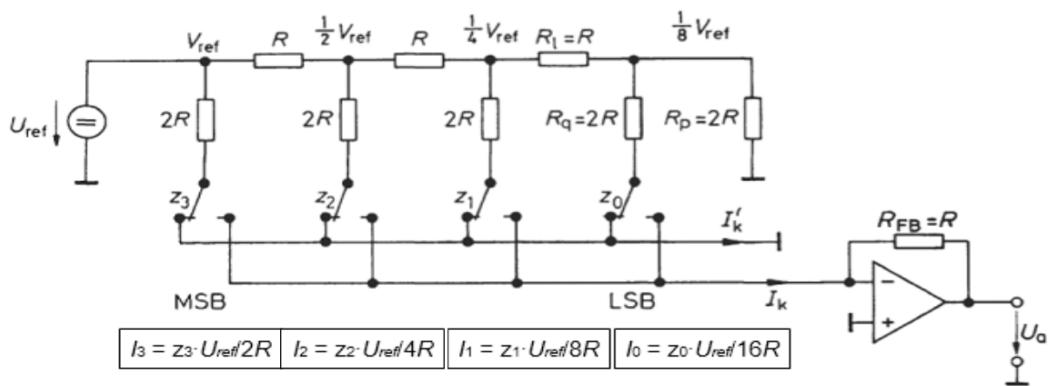
Because of the large number of switches required for high precision DACs, the parallel methods are used rarely. The counting methods become increasingly important, because here the pulse width modulator is an easy-to-integrate digital circuitry. If it operates at a frequency which is much higher than the sampling frequency (*oversampling*), only a simple **analog** lowpass filter is necessary.

Successive Approximation DAC



here: \$Z_{max} = 15\$

DAC with R-2R network (can be realized in CMOS technology)



here: \$Z_{max} = 15\$

$$U_a = -U_{ref} \frac{Z}{Z_{max} + 1}$$

Both SAR type DA converters shown above, are found in separate integrated circuits, often with a serial interface to a \$\mu\$C.

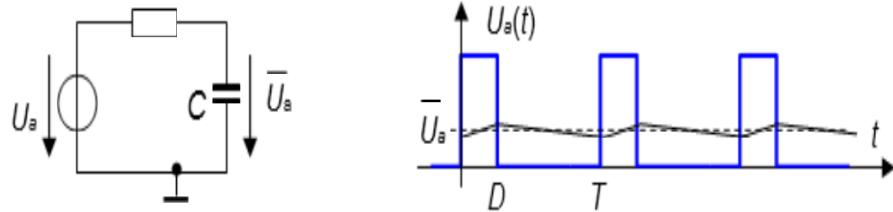
e.g. LTC1453 – Dual 12-Bit DACs in SO-8 package (SPI-Interface).

SAR type DACs are rarely found integrated with a microcontroller chip (exception: NXP-LPC1766 with integrated 10-bit SAR DAC). Many applications are realized with PWM DACs.

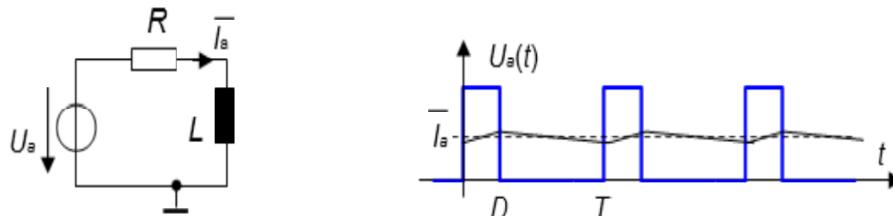
4.1.4 DA Converter (PWM)

Pulse-width modulation (PWM) is a modulation, where a physical quantity (e.g. an electric voltage) switches between two values at a constant frequency $f = T^{-1}$, with the duty cycle ratio $d = D / T$ of the modulated digital signal, with D as width of a pulse.

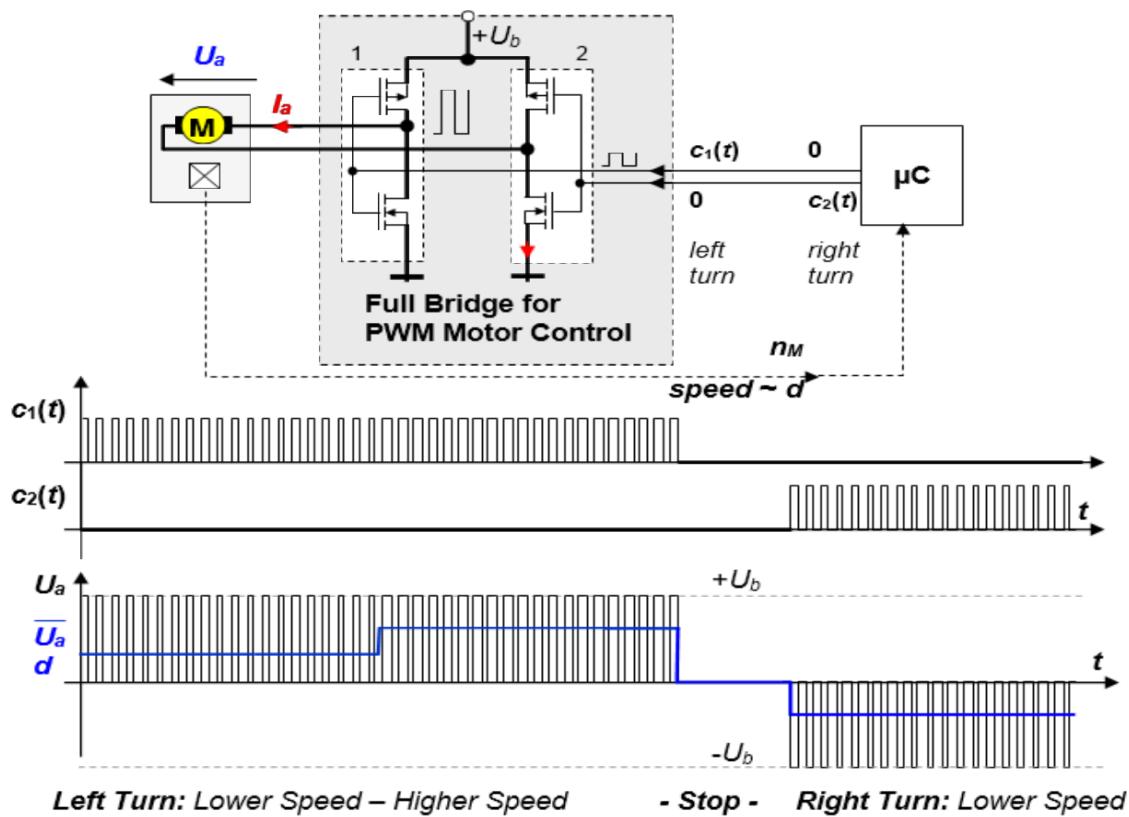
Example 1: PWM-DA converter with $U_a(t)$ generated at a μ C-Port with $T \ll R \bullet C$.



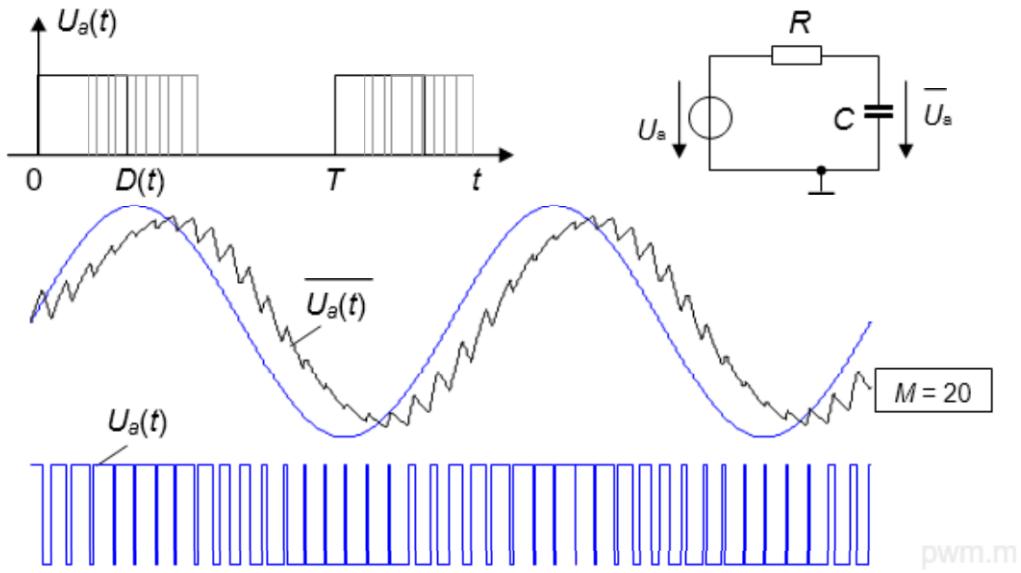
Example 2: PWM control of an R-L impedance with $T \ll L/R$.



Example 3: PWM Control with Direction Control of a DC-Motor Full Bridge.



Example 4: PWM control with a time function of the duty-cycle.



pwm.m

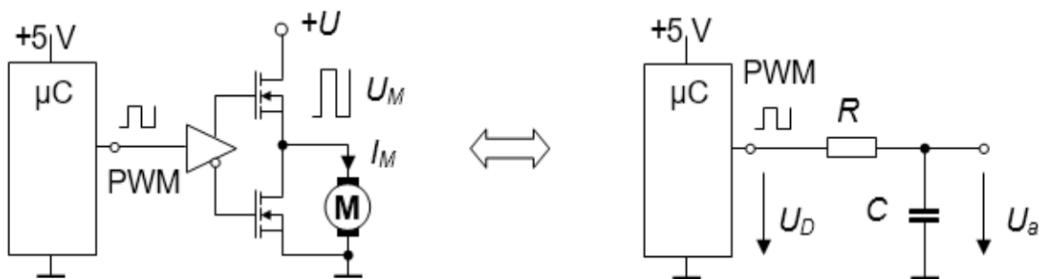
A PWM signal is demodulated, generally, by a low pass filter. The resulting demodulated technical quantity $d(t)$ is the short-term mean and thus, the mean height of the area under the modulated quantity $U_D(t)$. With an integration time constant $= R \cdot C$

$$U_a(t) \approx U_D(t) = \frac{1}{T} \int_0^T U_D(t) dt = \frac{D(t)}{T} \cdot \hat{u} = d(t) \cdot \hat{u} \quad (4.1)$$

the RC lowpass acts as an integrator, if $T < <$, or the PWM frequency is high compared to the bandwidth and the maximum frequency of the analog signal (oversampling ratio M).

→ The higher the time-constant related to the PWM periodic time, the smaller the "ripple"-amplitude.

In power electronics PWM is widely used, because the power dissipated into heat in the switch ($P_S = U_S \cdot I_S$) can be minimized close to 0, if the switch is either **fully closed** ($U_S = 0, I_S \neq 0$) or **fully open** ($U_S \neq 0, I_S = 0$).



A vivid example of PWM modulation is a pair of switches (realized as power MOSFETs), which is used to control the DC motor of a drive by constantly switching on and off.

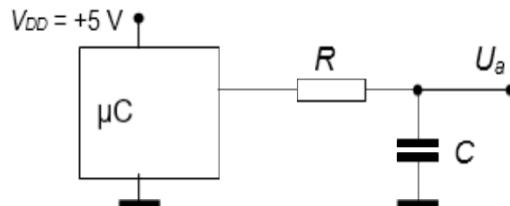
The higher the duty cycle ratio $d = \text{on-time} / (\text{off-time} + \text{on-time})$ of the switch, the higher the average speed. Due to lowpass characteristics of the motor, the motor acts as a lowpass filter with time-constant $= L_a/R_a \sim 20$ ms. Thus, speed cannot follow a 20 kHz PWM frequency (as a typical PWM frequency at 20 kHz (audio limit) is used, $T = 50 \mu\text{s}$). Since $T < <$, the motor speed is propor-

tional to the average duty cycle d .

Power loss with specified R_{on} of a half-bridge: $P_s = I_M^2 R_{on}$
Example: $R_{on} = 5 \Omega$, $I_M = 10 \text{ A}$ $P_s = 0.5 \text{ W}$
(very low power loss)

Example (from examination WS2010):

In a real-time program for a microcontroller a PWM DAC shall be realized for generating 2 different analog voltages U_{a1} and U_{a2} by means of a programmable PWM timer with period $T = 50 \mu\text{s}$ (20 kHz), and a 5 V digital output pin of the microcontroller, and an analog RC lowpass.



Which pulse width D_1 (or D_2) must be programmed to get $U_{a1} = 1 \text{ V}$ (to get $U_{a2} = 3 \text{ V}$) at the output of the RC-lowpass ?

$$\text{with (4.1): } \frac{D}{T} = \frac{U_a}{V_{DD}}$$

$$D_1 = 1 \text{ V} / 5 \text{ V} \cdot 50 \mu\text{s} = 10 \mu\text{s}.$$

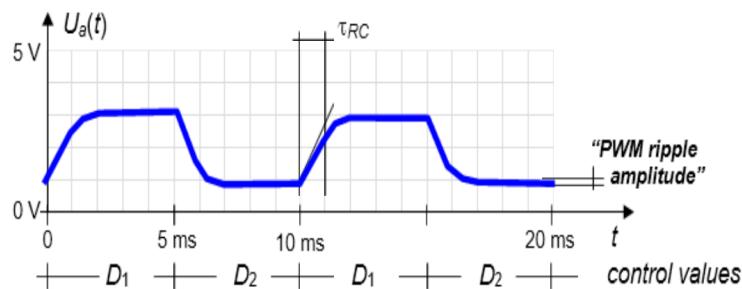
$$D_2 = 3 \text{ V} / 5 \text{ V} \cdot 50 \mu\text{s} = 30 \mu\text{s}.$$

What RC-time constant τ_{RC} is useful, if the voltage U_a is switched periodically between U_{a1} und U_{a2} at a periodic time $T_{on,off} = 10 \text{ ms}$?

$$\begin{aligned} T_{on,off} &= 5 \tau_{RC} & \rightarrow \tau_{RC} = T_{on,off}/5 = 2 \text{ ms} \\ \tau_{RC} &= R \cdot C >> T & \rightarrow 2 \text{ ms} >> 50 \mu\text{s} \text{ OK} \end{aligned}$$

→ The lower the time constant, the higher the **PWM-ripple amplitude** !

When switching the pulse durations D_1 (D_2) each 5 ms, one gets the signal $U_a(t)$



DA Converter (Delta-Sigma Method)

see 4.5.4

4.2 AD Converters

If an analog voltage shall be processed by a digital computer, that voltage must be translated into a corresponding number. This task is an analog-digital converter (ADC). With Z as an integer number, proportional to the analog input voltage U_e

$$Z = U_e / U_{LSB}$$

this is called a linear ADC, this type is most widespread type (for more specific applications, there are also logarithmic ADCs).

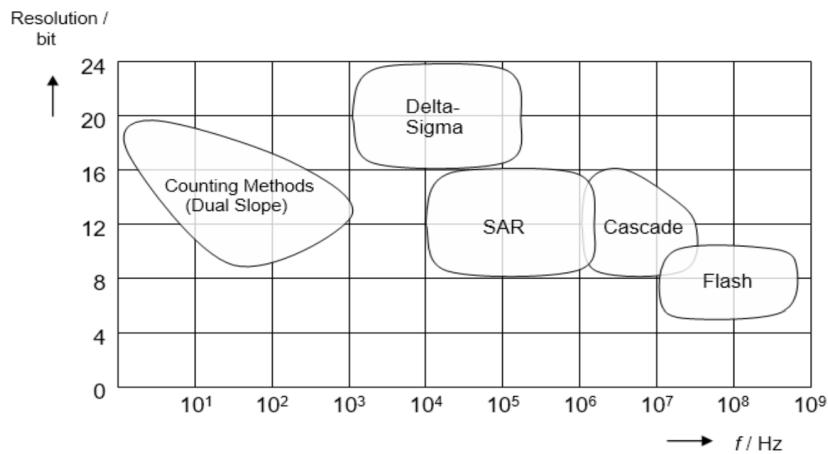
U_{LSB} is the voltage unit for the least significant bit (LSB), that is the voltage corresponding to $Z = 1$, it is derived from an **accurate reference voltage** U_{ref} .

One can basically distinguish four different methods:

1. the flash-method (parallel method), "word at a time"
2. the Successive-Approximation-method (SAR) "digit at a time"
3. counting methods "level at a time"
4. Sigma-Delta methods

There is a compromise between flash- and SAR-converters: **cascade converters** ("half-flash converters"), which are not presented here. These converters provide high resolution at high sampling frequencies [Tietze]. Cascade converters behave like flash ADCs from the perspective of software.

AD-Converters - Resolution and max. Sampling Frequencies



AD Converters (Flash-Method)

A flash ADC compares the input voltage simultaneously (in parallel) with n reference voltages and states, and compares the interval between the unknown input voltage is between. By this, one gets the full conversion in one step ("one word at a time"), very quickly.

However, the cost is very high, as is required for any possible number 1 .. Z_{max} a separate comparator is needed for a measuring range 0-255 steps (8-bit), 255 comparators.

Applications for flash ADCs: Video conversion, Oscilloscope, radar, ...

Advantages:

- very fast (conversion times of ~ 1 ns (8 bit, 600 MHz: TDA8718), or ~ 10 ns (10 bit, 75 MHz: AD 9060))
- due to the short aperture times, no sample & holds circuits necessary the ADC result is obtained with a simple read from a data register: $\mathbf{z} = \text{get}(\text{ADC})$

Disadvantages:

- no high resolutions available
- rel. expensive
- analog semiconductor process poorly implemented in CMOS technology

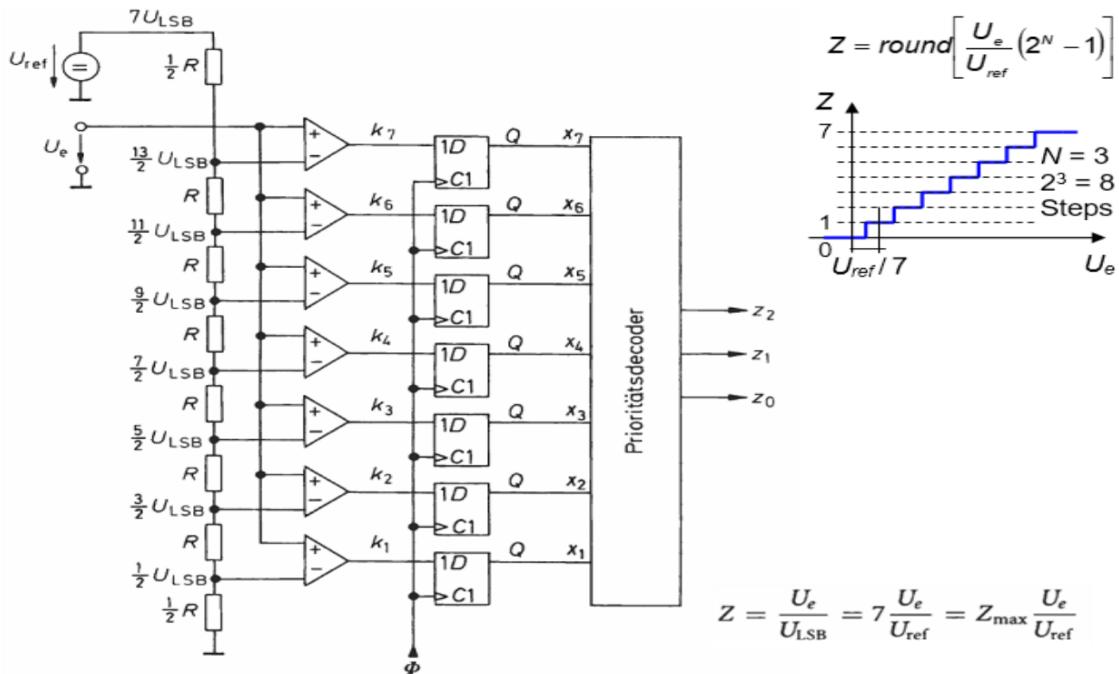


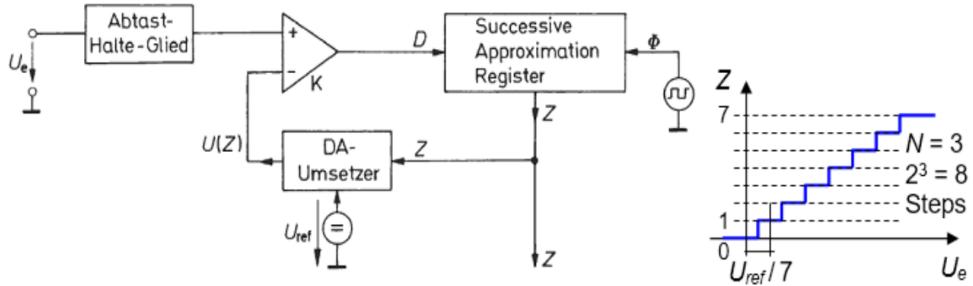
Figure 4.2.1: 3-Bit Flash AD Converter (Parallel Method)

AD Converter (Successive Approximation Method)

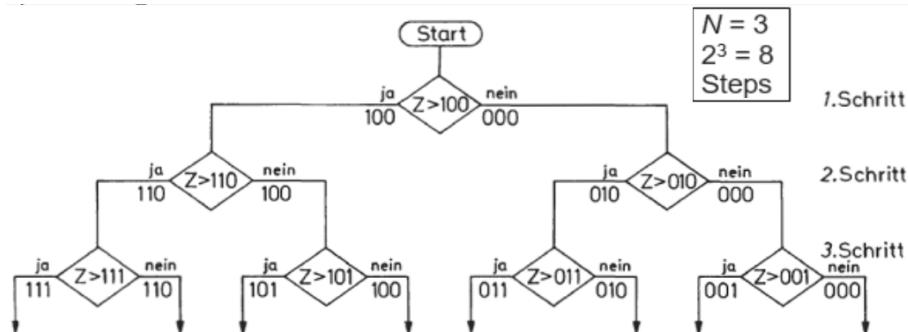
With the successive approximation register (SAR) method, the final result is not produced in one step, but rather a sequence of results of successive comparison steps, similar to weighing a mass using a balance. Each digit, having a different significance, of the binary representation of the corresponding binary number is determined separately and successively.

It starts with the highest bit and determines whether the input voltage is higher or lower than the reference voltage for the highest bit. If it is greater, the voltage for the highest bit is subtracted. The rest is compared with the next bit, So as many comparison steps are needed, as bits are necessary to represent the number.

$$Z = \text{round} \left[\frac{U_e}{U_{ref}} (2^N - 1) \right]$$



SAR-Converter with $Z_{max} = 7 \rightarrow 3$ steps. The SA-register is a digital sequential logic, which generates a voltage by a DAC a comparison voltage $U(Z)$, and by the following sequential algorithm:



Application: Microcontroller, fast metrology, radar, ...

Advantages:

- fast and precise (conversion times $\sim 1 \mu\text{s}$ (12 bit: TLV 5619))
- basically 1 comparator and digital circuitry necessary
→ realizable in CMOS, at reasonable cost, can be shrunk often integrated in microcontrollers

Disadvantages:

- due to the rel. long conversion times, sample & hold circuits are needed
→ the SAR-ADC must be started first, activating the S&H: `start_ADC()`
→ then, the ADC result is obtained with a read from a data register: `z = get(ADC)`

Example: ADC readout with AVR microcontroller

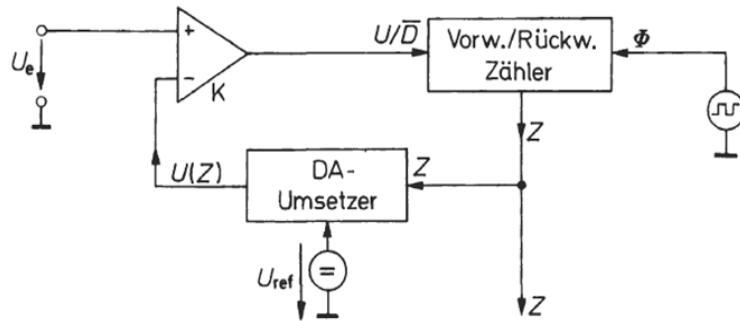
```

1 unsigned char tmp_lo, tmp_hi;
2
3 ADCSRA |= (1<<ADSC);           // start AD-conversion ...
4 while ( ADCSRA & (1<<ADSC) ) { // wait for AD to get finished
5 };
6 tmp_lo = ADCL;                  // read AD-data register
7 tmp_hi = ADCH;

```

AD Converter (Counting Method)

The simplest ADC method is the counting method. It is determined, how often some reference voltage has to be added, until the input voltage is reached. The number of steps is equal to the result. If the largest representable number is n , up to n steps are needed to get the result.



AD conversion by the counting method requires low electronic circuit complexity, but conversion time is significantly larger as with other methods. It is usually between 1 ms and 1 s. This is sufficient for slowly changing signals, such as in temperature measurement or for laboratory instruments, which are read by eye (digital voltmeter).

The "Dual-Slope"-counting method has the greatest importance, for which the best accuracy can be achieved with little effort.

Application: Lab Measurement Devices

Advantages:

- very precise (conversion times 1 ms ... 1 s
(HP, Keithley, Fluke – Lab devices: 5.5 decimal digits and more)

Disadvantages:

- due to the long conversion times, sample & hold circuits are needed

Example: Python Script for reading a measured voltage by GPIB:

```
see http://pyvisa.sourceforge.net
import visa
keithley = visa.instrument("GPIB::12")
print keithley.ask("*IDN?")
```



AD Converter (Delta-Sigma Method)

A delta-sigma-ADC consists of a delta-sigma modulator which produces the so-called **bitstream**, and a low pass filter in series.

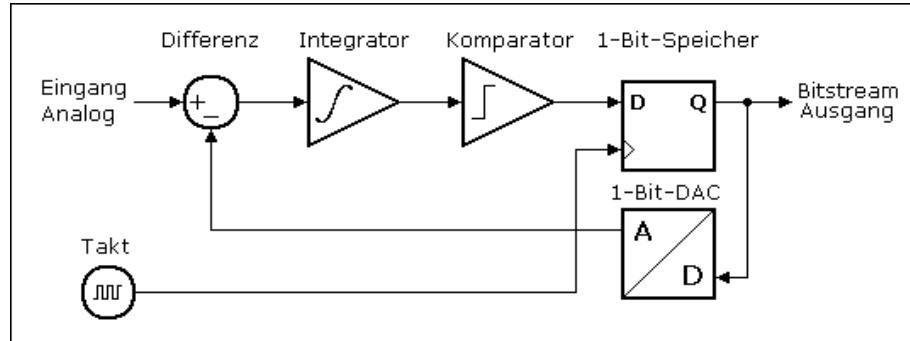


Figure 4.2.2: Delta-Sigma Modulator

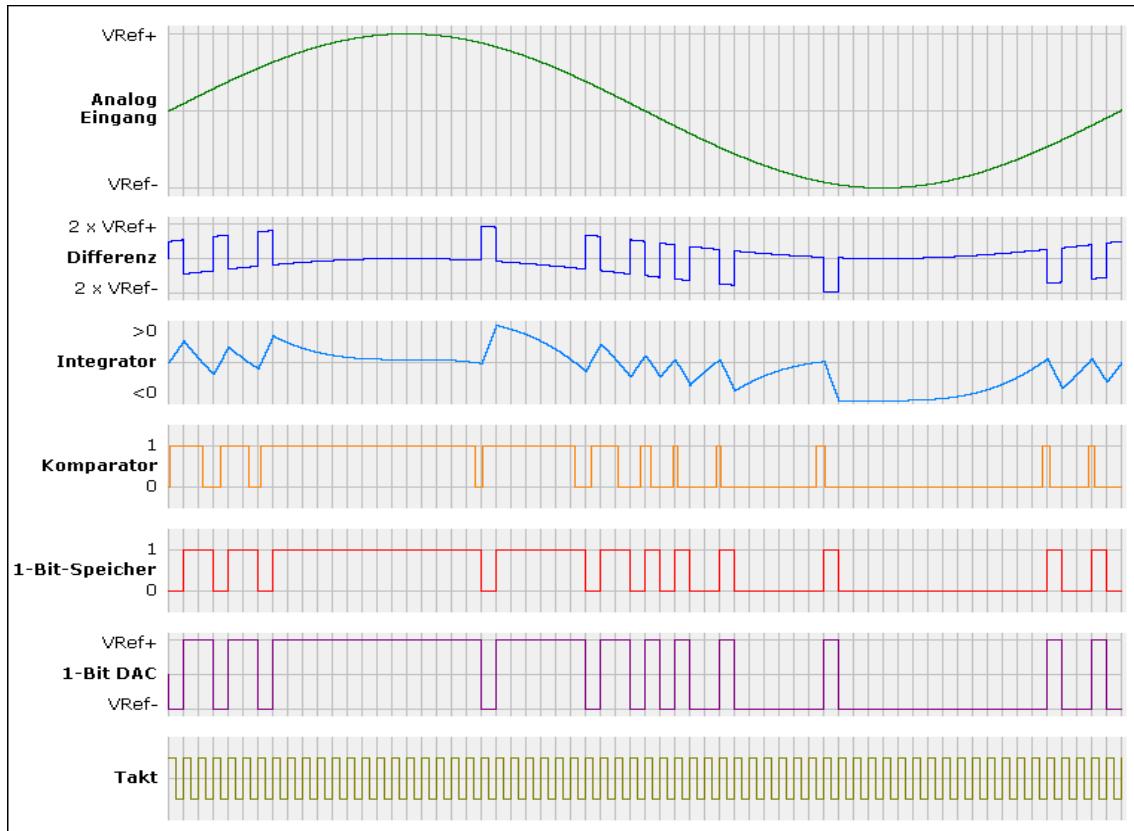


Figure 4.2.3: Signals with the Delta-Sigma Modulator

The output signal at the 1-bit memory contains the analog signal with a resolution of only one bit, but with considerably higher frequency components (in the picture factor 64). This is referred as oversampling M (here $M = 64$). Since the original analog signal is included in the 1-bit signal as a short-term average, it can be obtained by a low pass filter as a very good approximation to the analog signal (see also section 4.4.2 PWM DACs):

Delta Sigma ADC:



Figure 4.2.4

Thus, the digital low pass filter realizes the short-term average (interpolation) on the 1-bit square wave signal with a high oversampling factor M . Since both the input and the output of the filter are discrete-in-time and discrete-in-value, this low-pass filter is realized as a digital filter (in hardware, an easy task with standard CMOS semiconductor technology).

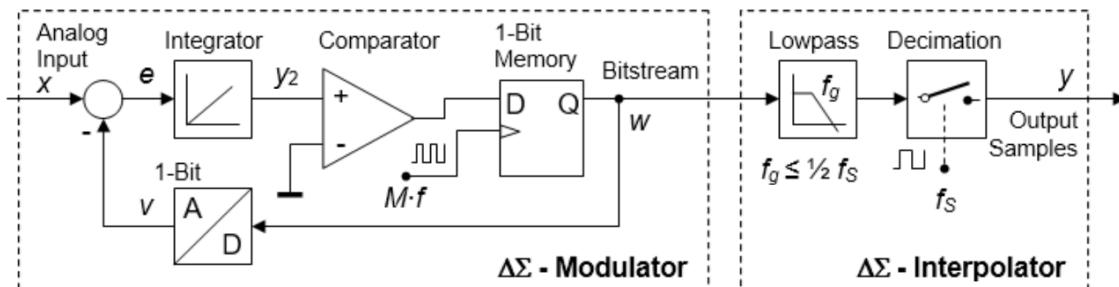
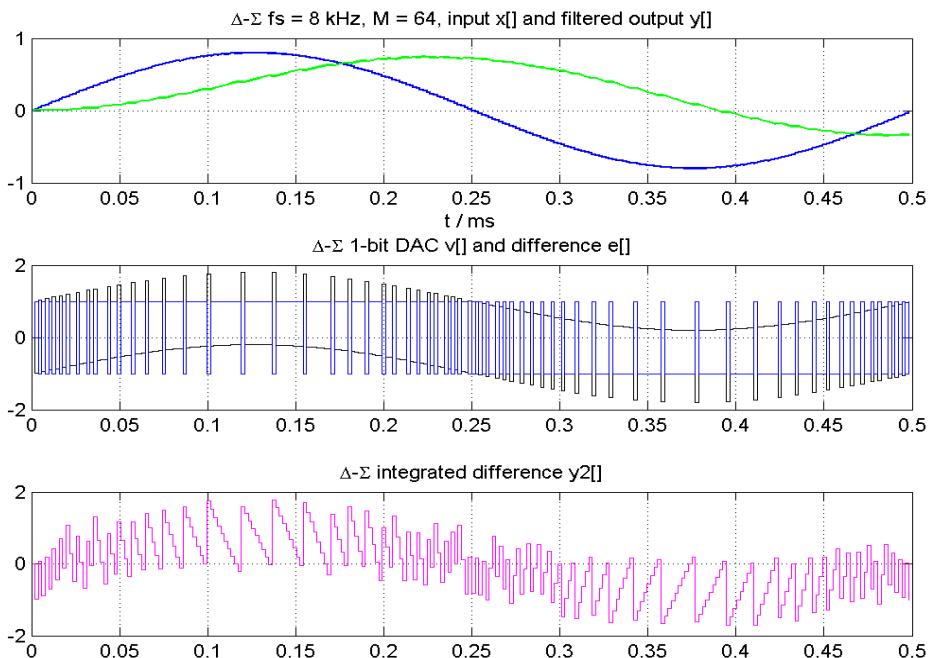


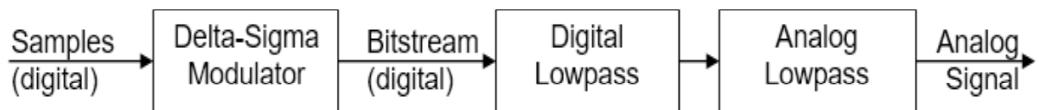
Figure 4.2.5



Waveforms in a -ADC with 8 kHz sampling frequency and 64-times oversampling. The low-pass filter is realized as a digital filter (usually as a FIR filter of high order).

A change of input- and output signal with the flowchart of the -ADC leads to a

Delta-Sigma DAC:



The realization of the delta-sigma modulator is purely digital, the low pass filter is usually a combination of a high order FIR digital filter and a (quite simple) analog lowpass filter. With a -DAC no decimator is needed.

Example: For audio Delta-Sigma converters (ADC and DAC) resolutions of 16 to 24 bit at sampling rates of 48 kHz are typical. The samples are processed in blocks of samples (typically 64-2048 values) using DMA (section 4.6) for memory transfer.

Advantages of Delta-Sigma Converters:

- fast, accurate AD- and DA- conversion possible
- no sample and hold circuits needed
- in addition to a single comparator only digital components
- needed → can be highly integrated in CMOS technology

Disadvantages of Delta-Sigma Converters:

- no accurate time reference, (because of the frequency-phase response of the low-pass filter necessary for process reasons)
- individual samples can not be captured, only blocks (typically 64-2048 samples)
- specification only possible in the frequency
- Main application: audio signal processing ("the ear works like a (magnitude) spectrum analyzer")

4.2.1 Direct Memory Access (DMA)

Peripherals are connected to the main processor by the processor- (system) bus. If data arrives with high frequency, the processor can not fetch the data word-by-word from the periphery effectively, due to the overhead. This can be avoided with transferring complete data blocks directly to the memory under control of the CPU:

Once started, the DMA controller, as a hardware component, reads data from the periphery in a loop and places them in memory (RAM), or vice versa.

MemoryPointer = RAM-Start address of the memory-block	1 (device)
TransferDirection = 1: RAM → I/O, 0: from I/O →	
Port = ID (address) of peripheral device	
Start DMA	
while BlockCount>0	
if TransferDirection	
1	0
value=	value = ReadIO(Port)
WriteIO(Port, value)	*MemoryPointer= value
MemoryPointer ++	
BlockCount --	
IRQ(DMA Ready)	
Unblock waiting Tasks	3 (CPU)

When DMA operation three phases:

1. **Initialization:** The CPU controls the bus (bus master), writes the DMA registers.
2. **DMA operation:** The DMA controller is bus master. It transfers data between peripherals and memory. In between, the bus is given free for some time intervals. This is repeated, until all data has been transferred (this phase is often referred to as "**cycle stealing**" because the DMA controller accesses the RAM via the data bus in competition with the CPU)
3. **End of DMA operation:** The DMA controller informs the CPU about the end by invoking an interrupt.

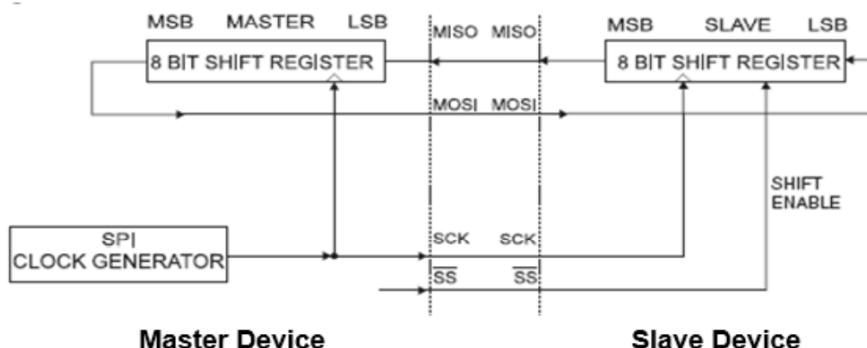
DMA is used to

- increase the transmission rate
- to relieve the CPU from trivial copy load

Applications: Fast peripherals, Video ADC / DAC, Audio Codecs, Hi-Speed Networks, mass storage.

4.2.2 Serial IO (Synchronous SPI)

The Serial Peripheral Interface (SPI) allows synchronous communication between a master device, which provides the clock (clock), and a slave device:



The clock frequency can be several MHz. A microcontroller can be configured as SPI master or slave device. The physical interface consists of three or four wires / pins:

4 interface pins:

- MOSI master out slave in
- MISO master in slave out
- SCK serial clock
- SS_n slave select

3 registers are needed:

- SPCR control register
- SPSR status register
- SPDR data register

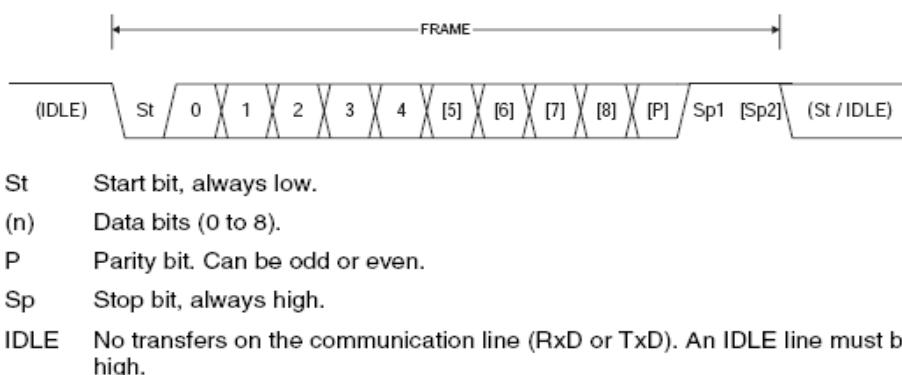
Code example:

```
1 // spi_init: initializes the SPI port on the AT mega128.
2 void spi_init(void){
3     DDRB = DDRB | 0x07;           // Turn on SS, MOSI, SCLK (SS is output)
4     SPCR |= (1<<SPEN) | (1<<MSTR); // spi enabled, master, low polarity, msb 1st
5     SPSR |= (1<<SPI2X);        // run at i/o clock (fosc/128 ... fosc/2) div 2
6 } //spi_init
7
8 // spi_send for AT mega128
9 void spi_send(uint8_t data[], uint8_t n){
10    for (int i=0; i < n; i++) {
11        SPDR = data[i];           // send data byte
12        while (bit_is_clear(SPSR,SPIF)) {} // wait till data is sent out
13    }
14 } //spi_send
```

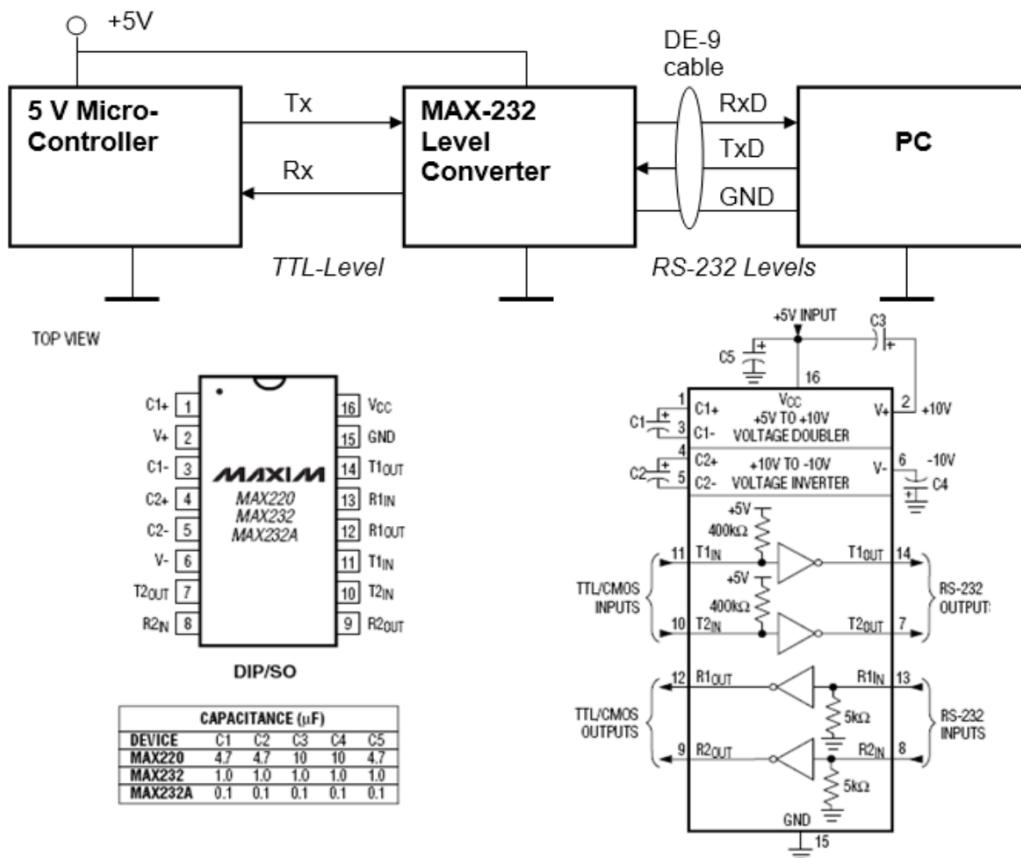
Example: LTC1453 - Dual 12-Bit SAR-DAC with SPI-Interface.

4.2.3 Serial IO (USART)

The Universal Synchronous / Asynchronous Receiver Transmitter (UART) interface provides both, synchronous data exchange (with clock, similar to the SPI) and asynchronous data exchange without clock. The interface is used primarily as a UART (asynchronous), to implement a RS-232 or RS-485 interface



Typical Application of an RS-232 Interface ($\pm 10V$) μ C \iff PC Communication



MAX220/MAX232/MAX232A Pin Configuration and Typical Operating Circuit

For synchronization, byte by byte (or by 7-, 6- or 5 bit wide words), always the beginning of the start bit is used. The baud rates can be set by the AT Mega UBRR register:

Baud Rate (bps)	$f_{osc} = 16.0000 \text{ MHz}$			
	$U2X = 0$		$U2X = 1$	
	UBRR	Error	UBRR	Error
2400	416	-0.1%	832	0.0%
4800	207	0.2%	416	-0.1%
9600	103	0.2%	207	0.2%
14.4k	68	0.6%	138	-0.1%
19.2k	51	0.2%	103	0.2%
28.8k	34	-0.8%	68	0.6%
38.4k	25	0.2%	51	0.2%
57.6k	16	2.1%	34	-0.8%
76.8k	12	0.2%	25	0.2%
115.2k	8	-3.5%	16	2.1%
230.4k	3	8.5%	8	-3.5%
250k	3	0.0%	7	0.0%
0.5M	1	0.0%	3	0.0%
1M	0	0.0%	1	0.0%
Max (1)	1 Mbps		2 Mbps	

1. $UBRR = 0$, Error = 0.0%

Code Examples:

```

1 void uart_init(){
2     //rx and tx enable, 8 bit characters
3     UCSRB |= (1<<RXEN0) | (1<<TXEN0);
4     //async operation, no parity, one stop bit, 8-bit characters
5     UCSRC |= (1<<UCSZ01) | (1<<UCSZ00);
6     // UBRR0H=0x00; //set to 9600 baud
7     // UBRR0L=0x67;

```

```

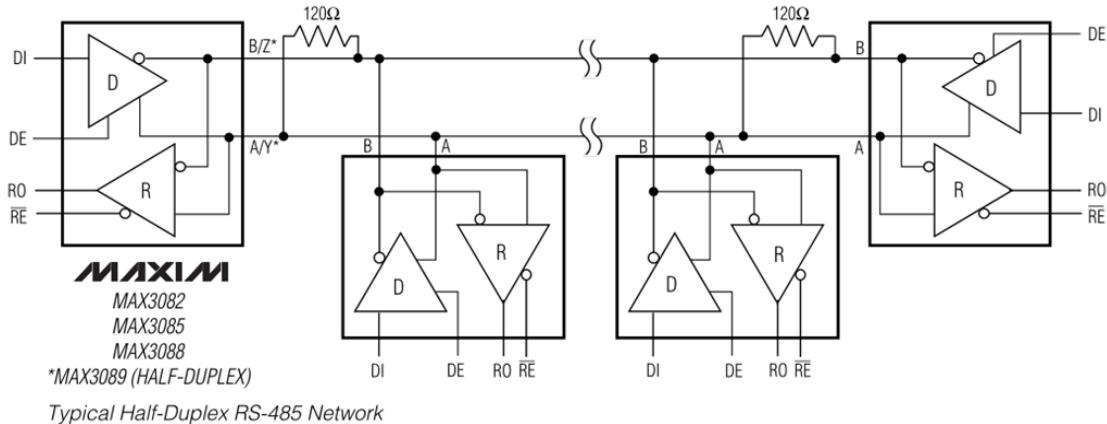
8 UBRR0H=0x00; //set to 115200 baud
9 UBRR0L=0x08;
10 UCSRA |= (1 << U2X); // double speed mode
11 }
12
13 // Takes a character and sends it to USART0
14 void uart_putc(char data){
15     while (!(UCSR0A & (1<<UDRE))); // Wait for previous transmission
16     UDR0 = data; // Send data byte
17     while (!(UCSR0A & (1<<UDRE))); // Wait for previous transmission
18 }
19
20 // ..Modified to not block indefinitely in the case of a lost byte
21 char uart_getc(void) {
22     uint16_t timer = 0;
23     while !(UCSR0A & (1<<RXC0))) {
24         timer++;
25         if(timer >= 16000) return 0; // ..not block indefinitely
26     } // Wait for byte to arrive
27     return UDR0;

```

Higher Baudrates than 115.2 kBd are possible with most microcontrollers (see table above, up to 2 MBd) possible, if the physical layer is capable of higher baudrates.

This is true for the RS-485 interface, where the UART is used in connection with a highspeed-transceiver for tristed-pair lines for up to 12 MBd:

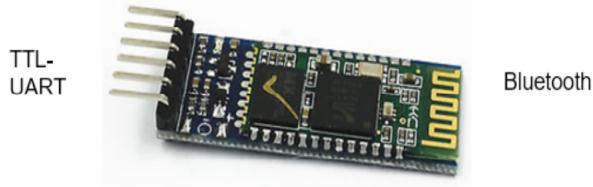
Half-Duplex RS-485 Network with MAX488-MAX491



For short line lengths USB transceivers for TTL levels (0 .. 3.3V oder 5V) are available (z.B. FT232), which can be operated at baudrates up to 2 MBd.



Bluetooth-modules offer a "virtual serial interface" with baudrates up to 2 MBd either:



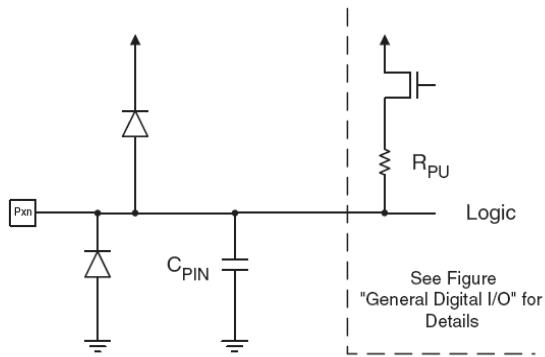
All these UART variants can be controlled via the "Streaming API" of Unix, Windows, ..

```
1 // open some com port ...
2 hCom = CreateFile("com9", GENERIC_READ | GENERIC_WRITE, 0, NULL, ... );
3 ...
4     ReadFile(hCom, buffer, sizeof(buffer), &read, NULL);           // read from com
5     port
6 ...
7     WriteFile(hCom, buffer, sizeof(buffer), &written, NULL); // write to com port
8 ...
9 CloseHandle(hCom);
```

4.2.4 Parallel IO (Ports)

A pin of a microcontroller is the only **path to the process**. As only a limited number of pins is available, the pins are designed to be as universal as possible, so that a pin can be used for various applications, such, it can be used either as input or output.

As the **technology of the microcontroller (CMOS)** is sensitive to electrostatic discharge (ESD) and the associated currents, there is always a **protective circuit** (diode) connected directly to the port pins:

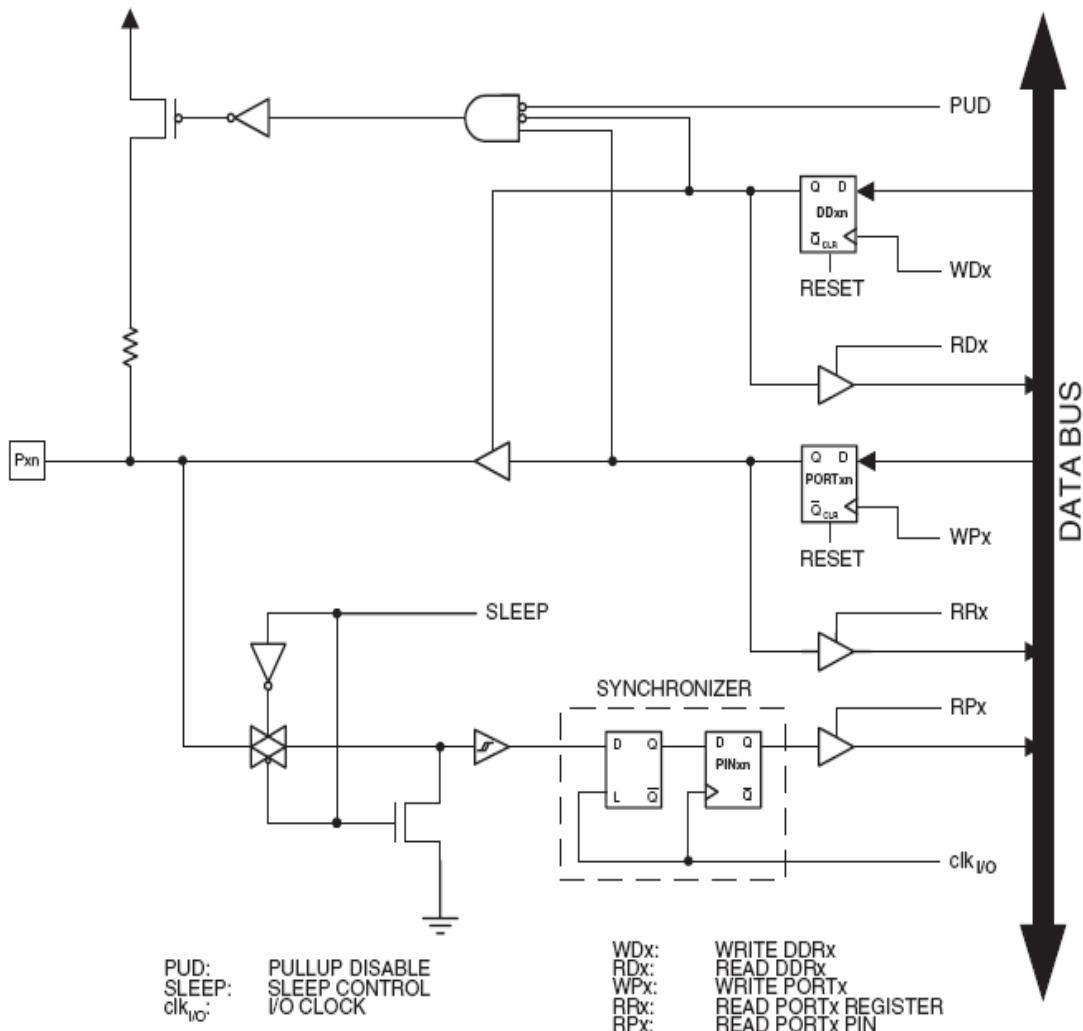


It is important to note that

1. the protection diodes are conductive for over voltage ($V_{CC} + 0.6V$) or low voltage (-0.6 V) and they are limited with their maximum current rating.
2. the risk of damage by ESD at a high-impedance input port is greatest.
3. inputs with a selectable pull-up resistor can be provided.

4. IO ports can be usually programmed for both directions. After a reset all pins are inputs **by default**, and thus are **high impedance** (3-state) Later, at software initialization the port can be set as output (by setting bits in the Data Direction Register (DDR), or, a pull-up resistor can be switched active.

AVR Port Architecture:



All ports have

- Bit selectable pull-up resistors
- Bit-selectable tri-state outputs
- Schmitt trigger input buffer
- Symmetrical DC Drivers
- Are synchronized with the internal clock

All ports have "read-modify-write" capability → one can re-program or change the value of each port bit independently of the others.

Port and Pin Control Register ($x = [A, B, C, D, E, F, G], [0 \dots 7]$):

- DDRx data direction register
- PORTx output register
- PINx input register

All port registers can be read, a write in the PINx register has no effect.

Example: Initialization of Port B: bit 0 and bit 1 as outputs, bits 2..7 as inputs with active pullup resistor:

```

1 // PB1: output for LED0 (1 = ON)
2 // PB0: output for LED1 (1 = ON)
3 #define LED0 (1<< PB0)
4 #define LED1 (1<< PB1)
5 #define DDRB_MASK (LED0) | (LED1)
6
7 // no open inputs PB2..PB7: activate internal Pullup
8 #define UNUSED_MASK (0xFC)
9 ...
10 #include <avr\io.h> // AVR register and constants definitions
11 ...
12 init() {
13     // -----
14     // Pin mappings
15     // DDRx PORTx Function of the Port pin
16     // 0 0 Input (high impedance)
17     // 0 1 Input with internal pullup
18     // 1 0 Output: LOW
19     // 1 1 Output: HIGH
20     // -----
21     DDRB = DDRB_MASK;           // set port pins to output (and input)
22     PORTB = UNUSED_MASK;       // set pullups for unused input port pins
23     // -----
24     ...
25     sei();                     // enable interrupts
26 }
27
28 void inline set_LED0(int bit) {
29     // driver function for p
30     if (bit)
31         PORTB |= LED0;    // on
32     else
33         PORTB &= ~LED0;   // off
34 }
```

Ports can be configured to issue an interrupt, if a certain slope (rising/falling) is applied.

4.2.5 Interrupts

When an interrupt occurs, execution jumps to a specific location in program memory, these locations (addresses) are called interrupt vectors, which can be used for assigning a user function to a certain interrupt. The interrupt assignments are fixed and can not be changed,

1	0x000	RESET	Reset after power-on
2	0x001	INT0	External Interrupt 0
3	0x002	INT1	External Interrupt 1
4	0x003	TIMER2_COMP	Timer/Counter2 Compare Match
5	0x004	TIMER2_OVF	Timer/Counter2 Overflow
6	0x005	TIMER1_CAPT	Timer/Counter1 Capture Event
7	0x006	TIMER1_COMPA	Timer/Counter1 Compare Match A
8	0x007	TIMER1_COMPB	Timer/Counter1 Compare Match B
9	0x008	TIMER1_OVF	Timer/Counter1 Overflow
10	0x009	TIMER0_OVF	Timer/Counter0 Overflow
11	0x00A	SPI_STC	SPI-transmission finished
12	0x00B	USART_RX	USART-receive finished
13	0x00C	USART_UDRE	USART-data register empty
14	0x00D	USART_TX	USART-transmit finished
15	0x00E	ADC	AD-conversion finished
16	0x00F	EE_RDY	EEPROM ready
17	0x010	ANA_COMP	analog comparator
18	0x011	TWI	two-wire Interface
19	0x012	SPM_RDY	Store Program Memory Ready

Example: External Interrupt with the asynchronous programming example

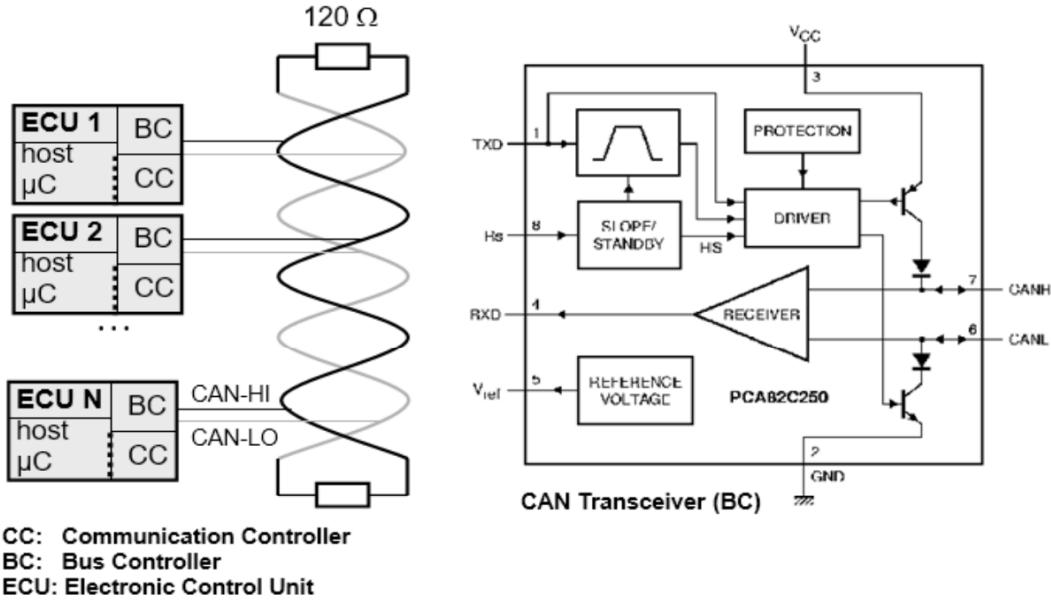
```

1 void inputISR_INT0(void) // called for any slope of INP \rightarrow INT0
2 {
3     OUT = 1;
4     // optional: some delay here
5     OUT = 0;
6 }
7
8 void main()
9 {
10    // init AVR MCUCR Register:
11    // external interrupt INT0 for both slopes
12    MCUCR.ISC00 = 1;
13    MCUCR.ISC01 = 0;
14
15    INT0addr = inputISR_INT0; // set INT0 interrupt vector
16
17    while (1); // forever
18 }
```

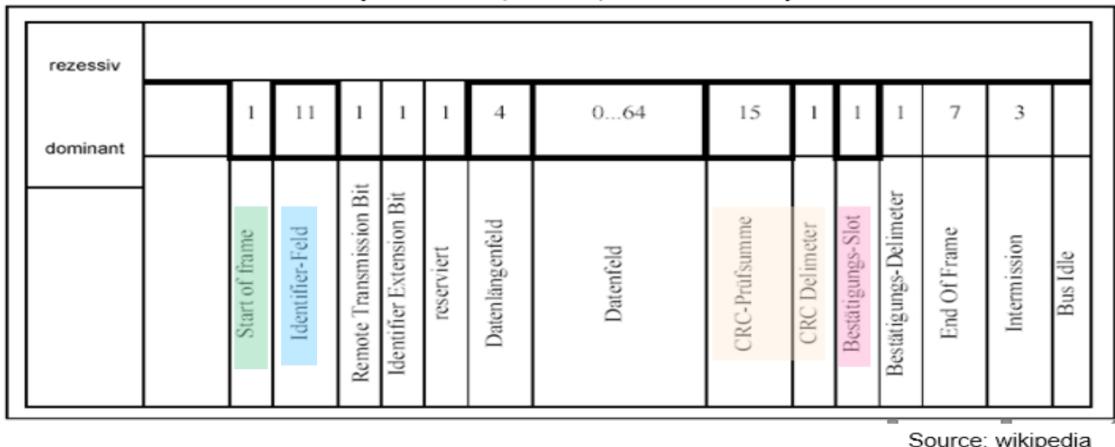
→ behavior of a slope-triggered flip-flop

4.2.6 CAN Networking

CAN-Networks are widespread in both automotive and industrial applications. With the well-proven twisted pair physical layer in line (bus) structure datarates up to 1 MBit/s are possible. The CAN-typical bus arbitration method uses prioritized message-IDs, using the dominance of a 0-bit over a 1-bit to avoid collisions.



CAN Standard Data Frame (CAN 2.0 A, BCAN, “Basic CAN”)

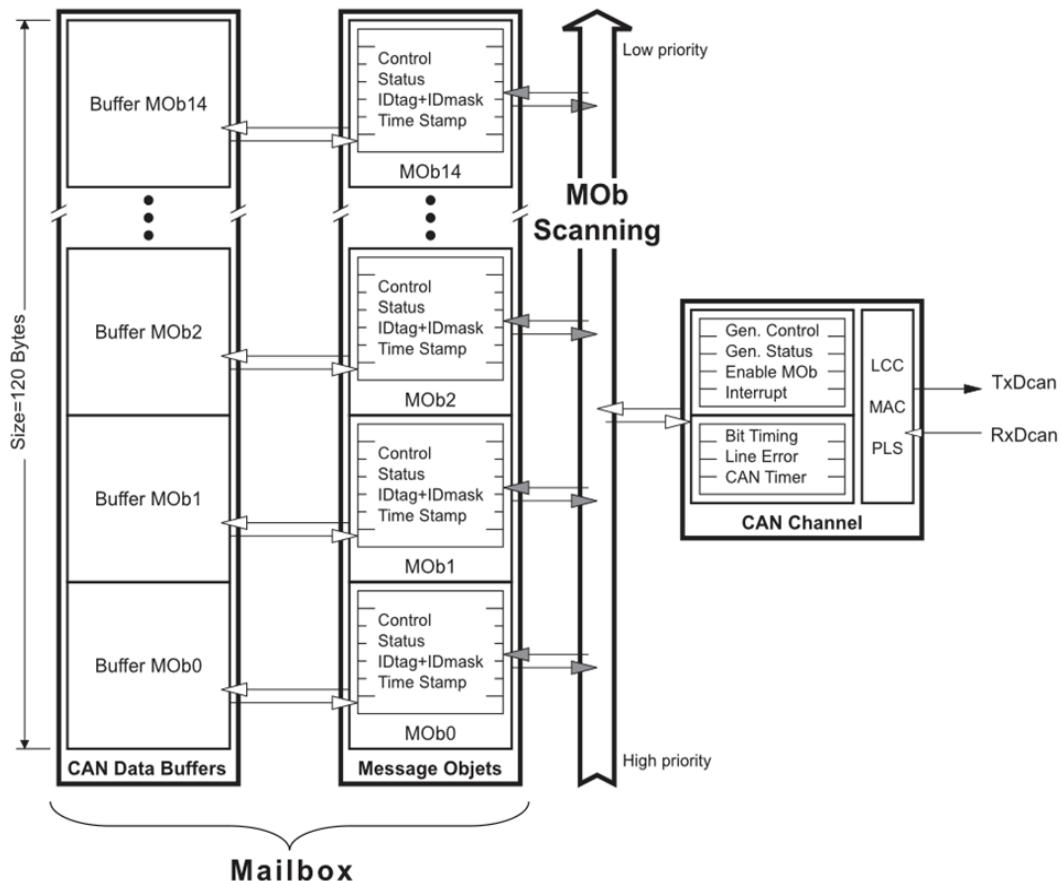


1. Synchronization by SOF
2. NRZ Line-Encoding with Bitstuffing after 5 equal Bits
3. Bus Arbitration using dominant 0s in Identifier Field (11 or 29 bits)
4. Error Checking: bit-, format-, CRC-, ACK-, Bitstuffing Monitoring

The CAN communication controller embedded as peripheral device into a microcontroller can be a **basic-** or a **full-CAN** controller. A basic CAN controller typically has 1 separate receive/transmit (RX-/TX-) buffer, and a mask can be set for message-ID filtering.

A **full-CAN** microcontroller provides the whole hardware for convenient acceptance filtering and message management for up to 16 message buffers. For each message to be transmitted or received the controller contains an MOb, in which all information regarding the message (e.g. identifier, data bytes etc.) are stored.

Example: AVR90CAN Full-CAN Controller.



During the initialization of the peripheral, the application defines which messages are to be sent and which are to be received. Only if the CAN controller receives a message whose identifier matches with one of the identifiers of the programmed (receive-) message objects the message is stored and the application is informed by interrupt.

Another feature is, that incoming remote frames can be answered automatically with the corresponding data frame. In this way, the CPU load is strongly reduced compared to a basic-CAN solution.

Example: CAN Workshop CAN Application (FreeRTOS)// @brief CAN message object structure

```

1 // @brief CAN message object structure
2 typedef struct {
3     uint32_t id;    // ID of message, if bit 30 is set then this is extended
4     frame */
5     uint32_t dlc;   // Message data length
6     uint8_t data[8]; // Message data
7 } message_object;
8
9 // messages for CAN RX, TX
10 message_object send_obj;

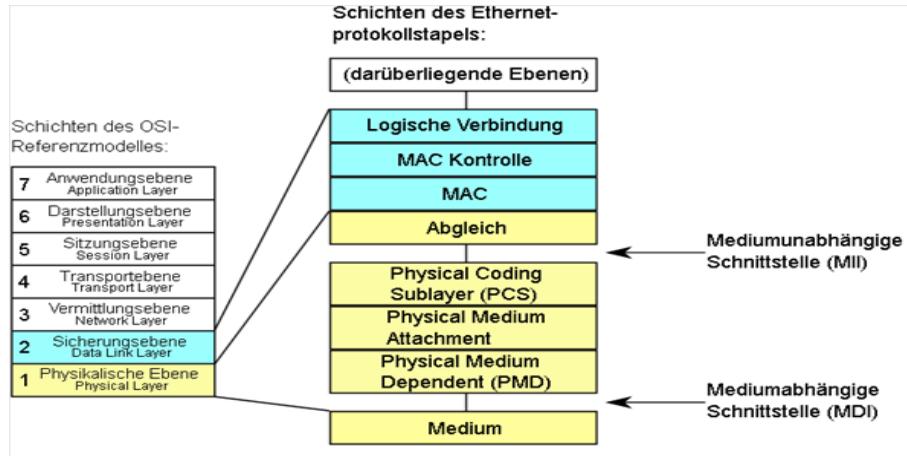
```

```

10 message_object rx_msg = {0, 0, 0,0,0,0,0,0,0,0,0,0};
11
12 // -----
13 void vCAN_Task(void * pvParameters)      // CAN TX task: periodic, 20 ms
14 {
15     portTickType xLastWakeTime = xTaskGetTickCount();
16
17     // can_init();
18
19     while(1) {
20         vTaskDelayUntil(&xLastWakeTime, txcycle/portTICK_RATE_MS ); // 20 ms
21
22         if (app_struct.team_nr) {                                // with team_nr set to > 0
23             send_obj.id = CCAN_TX_MSG_ID + app_struct.team_nr;
24             send_obj.dlc = 3;                                     // 3 data bytes
25             send_obj.data[0] = app_struct._d[6];                  // joystick state
26             send_obj.data[1] = (uint8_t)app_struct._d[7];        // poti LSB
27             send_obj.data[2] = (uint8_t)(app_struct._d[7]>>8); // poti MSB
28             can_send(&send_obj, 0);
29         }
30     }
31 }
32
33 // -----
34 uint32_t can_int, can_stat;
35 // -----
36 void CAN0_IRQHandler(void)           // CAN RX interrupt service routine:
37 {
38     message_object msg_buf;
39     uint32_t /*can_int, can_stat,*/ i;
40     while ( (can_int = Chip_CCAN_GetIntID(LPC_C_CAN0)) != 0 ) {
41         if (can_int == CCAN_STATUS_INT) {
42             can_stat = Chip_CCAN_GetStatus(LPC_C_CAN0);
43             // TODO with error or TXOK, RXOK
44             if (can_stat & CCAN_STAT_EPASS) {
45                 DEBUGOUT("Passive error\r\n");
46                 return;
47             }
48             if (can_stat & CCAN_STAT_EWARN) {
49                 DEBUGOUT("Warning !!!\r\n");
50                 return;
51             }
52             if (can_stat & CCAN_STAT_BOFF) {
53                 DEBUGOUT("CAN bus is off\r\n");
54                 return;
55             }
56             Chip_CCAN_ClearStatus(LPC_C_CAN0, CCAN_STAT_TXOK);
57             Chip_CCAN_ClearStatus(LPC_C_CAN0, CCAN_STAT_RXOK);
58         }
59         switch (msg_buf.id) {
60
61             case CCAN_RX_MSG_ID_LED:          // 20 <team_nr> [<mask>]
62                 Chip_CCAN_ClearIntPend(LPC_C_CAN0, can_int, CCAN_RX_DIR);
63                 ...

```

4.2.7 Ethernet



MII-Interface

On the hardware side, the Media Independent Interface (**MII**) is a standard interface used to connect a Fast Ethernet (i.e. 100 Mbit/s) MAC-block to a PHY chip (see 3.3.7).

While formerly the MII was used as a physical pluggable connector, it is used mainly to connect the MAC layer to a PHY chip. Being media independent means that any of several different types of PHY devices for connecting to different media (i.e. twisted pair copper, fiber optic, etc.). The equivalents of **MII** for other speeds are **AUI** (for 10-Base Ethernet), **GMII** (for Gigabit Ethernet), and **XGMII** (for 10 Gigabit Ethernet).

RAW-Socket Interface

On the software side, the socket interface (see section 2.3.9) is most popular and well established on almost any platform for TCP/IP and UDP/IP communications. For bypassing transport and network layers 4 and 3, socket interface provides a "Raw-Socket" interface allowing to send/receive Ethernet frames directly.

Example: Programming with "RAW-Sockets" (=Ethernet Frames 10.2.4), on OSI-layer 2

	Bytes	8	6	6	2	0-1500	0-46	4	
(a)		Preamble	Destination address	Source address	Type	Data ↓↓	Pad	Check-sum	
(b)		Preamble	S o F	Destination address	Source address	Length ↓↓	Data ↓↓	Pad	Check-sum

1. Create a RAW ethernet socket ...

```
1 #include <sys/socket.h>
2 #include <linux/if_packet.h>
3 #include <linux/if_ether.h>
4 #include <linux/if_arp.h>
5 ...
6 int s; // socketdescriptor
7
8 s = socket(AF_PACKET, SOCK_RAW, htons(ETH_P_ALL));
9 if (s == -1) { errorhandling ... }
```

2. Send a RAW ethernet frame ...

```
1 #define ETH_FRAME_LEN 1518          // max. frame, without preamble
2 ...
3 // target address
4 struct sockaddr_ll socket_address;
5
6 // buffer for ethernet frame
7 void* buffer = (void*)malloc(ETH_FRAME_LEN);
8
9 // pointer to ethernet header
10 unsigned char* etherhead = buffer;
11
12 // userdata in ethernet frame
13 unsigned char* data = buffer + 14;
14
15 // another pointer to ethernet header
16 struct ethhdr *eh = (struct ethhdr *)etherhead;
17
18 int send_result = 0;
19
20 // our (source) MAC address
21 unsigned char src_mac[6] = {0x00, 0x01, 0x02, 0xFA, 0x70, 0xAA};
22
23 // other host (destination) MAC address
24 unsigned char dest_mac[6] = {0x00, 0x04, 0x75, 0xC8, 0x28, 0xE5};
25
26 // RAW communication
27 socket_address.sll_family = PF_PACKET;
28 ...
29
30 // MAC - begin
31 socket_address.sll_addr[0] = dest_mac[0];
32 socket_address.sll_addr[1] = dest_mac[1];
33 socket_address.sll_addr[2] = dest_mac[2];
34 socket_address.sll_addr[3] = dest_mac[3];
35 socket_address.sll_addr[4] = dest_mac[4];
36 socket_address.sll_addr[5] = dest_mac[5];
37 // MAC - end
38 socket_address.sll_addr[6] = 0x00; // not used
39 socket_address.sll_addr[7] = 0x00; // not used
40
41
42 // set the frame header
43 memcpy((void*)buffer, (void*)dest_mac, ETH_ALEN);
44 memcpy((void*)(buffer+ETH_ALEN), (void*)src_mac, ETH_ALEN);
45 eh->h_proto = 0x00;
46 // fill the frame with random data
```

```

47 for (j = 0; j < 1500; j++) {
48     data[j] = (unsigned char) (256 * rand() / RAND_MAX);
49 }
50
51 // send the packet
52 send_result = sendto(s, buffer, ETH_FRAME_LEN, 0,
53                      (struct sockaddr*)&socket_address, sizeof(socket_address));
54 if (send_result == -1) { errorhandling... }

```

3. Receive a RAW ethernet frame ...

```

1 void* buffer = (void*)malloc(ETH_FRAME_LEN); // Buffer for ethernet frame
2 int length = 0; // length of the received frame
3 ...
4 length = recvfrom(s, buffer, ETH_FRAME_LEN, 0, NULL, NULL); // blocking ...
5 if (length == -1) { errorhandling .... }

```

Chapter 5

Real-Time System Development

Today, full compliance with all requirements can not be guaranteed. Thus, effective use of development methods and tools can reduce the probability for the occurrence of damage due to errors (software bugs) in a product.

In many products specific real-time operating systems (RTOSes) are used, which in comparison to universal OSes, having limited functionality, but much more predictable behaviour, which is the key issue with real-time systems.

Despite the use of RTOSes, the temporal system behavior is difficult to predict, especially with high CPU loads. Thus, one approach is to utilize the processor(s) only 30 .. 50%..

Host-Target Environment

Embedded systems development usually takes place in host-target environments, mainly with a PC as a host on which the development environment is installed, and a binary file to download to the target system. Since with C programming language programming close-to-the-hardware is relatively simple and required, C dominates the embedded much more than any other programming language (but still, for computationally intensive algorithms assembly language is used). With applications using powerful computer systems, there are other programming languages, for example in avionics systems Ada is used.

Software Tests

A software test is a test during software development to measure the functionality of a software to the requirements and their quality, and to identify software errors.

1. dynamic measures that require implementation of the software (e.g. test coverage)
2. static measures that can do without an implementation of the software (e.g. code reviews)

There are several definitions for software testing:

ANSI/IEEE Std. 610.12-1990 defines test as the process of operating a system or component under specified conditions, observing or recording the results and making an evaluation of some aspects of the system or component.

Denert [1], defines test as "the verifiable and repeatable evidence of the correctness of a software building module relative to pre-established requirements".

Aim

The aim of software testing is to detect errors !

But: to demonstrate that there are no errors, is not purpose of software testing !

If in a test no errors are found, this never proves that the program is correct. The correctness of the program can't be proven by testing (except in trivial cases) !

The reason for this is that for a prove of correctness, all combinations of all possible values of the input data had to be tested:

A simple realistic example:

→ 3 input variables, each 16 bit long leads to $N = 2^{3 \cdot 16} = 2^{48} \sim 10^{16}$ possible combinations, with 1 s time for testing each combination, a complete test would take 8.9 Mio years. The number of combinations with real programs is too large for complete tests.

For this reason, different test strategies / concepts exist to deal with the question, how to achieve a maximum test coverage with the lowest possible number of test cases.

The results of software tests help to assess the true quality of the software.

Test Planning

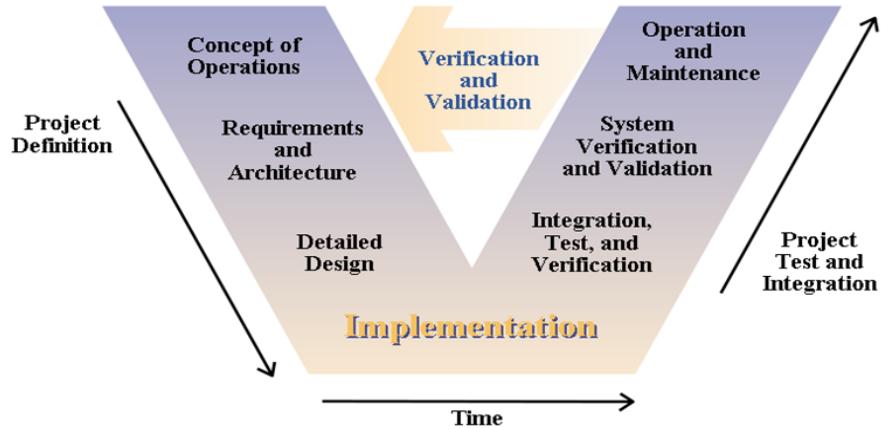
1. What **requirements** shall be tested in what depth ?
2. The number and the complexity of the test cases must be sufficient.
3. Tests should be done at software development, ideally by the developers at function/module level.

The test design takes place in parallel to software development.

Requirements Engineering (RM)

1. The purpose of Requirements Management is to ensure that **complex products and systems** can be developed **in time**, with the right **quality**, at the right **costs** !
2. Requirements Management means, that **processes** are defined and implemented for the tracing and documenting the complete development process: "intelligibility, clarity, traceability, consistency, completeness, testability". This **processes** finally set the basis for defining the test cases !
3. RM is well established in automotive industry ISO/IEC 15504 (**SPICE**). RM is closely connected to version- and configuration-management. RM is supported by database tools (e.g. **DOORS (IBM/Rational)**)

V-Model of Testing



The V-Model (or VEE model) of a systems engineering process.

The classification of the test levels follows the timely progress of the project. In general, there are 4 test levels or test phases in the V-model

1. unit test,
2. integration test,
3. system test and
4. acceptance test.

During development at each stage is tested against the corresponding specification.

1. Unit Test

A unit test (component test) is a test on the deepest level. There, individual components are tested for proper functionality (hardware and software). Frequently, the software module test is carried out by the software developer, the test objects are single defined modules, functions or classes.

2. Integration Test

An integration test (or interaction test) tests the compliance of independent components. This can be hardware and software components, or integration of all tasks running with a common task schedule.

3. System Test

With a system test the whole system is tested against the total requirements (functional and non functional requirements tested). Usually the test is done with automatic test environment using test data to simulate the final environment (e.g. squeeze force tests with a door ECU with a software anti-squeeze function).

4. Acceptance Test

An acceptance test, or user acceptance test (UAT) is a test of the product by the customer. Often, acceptance tests are required for billing.

In an acceptance test, the black-box method is used, i.e. the customer does not consider the code of the

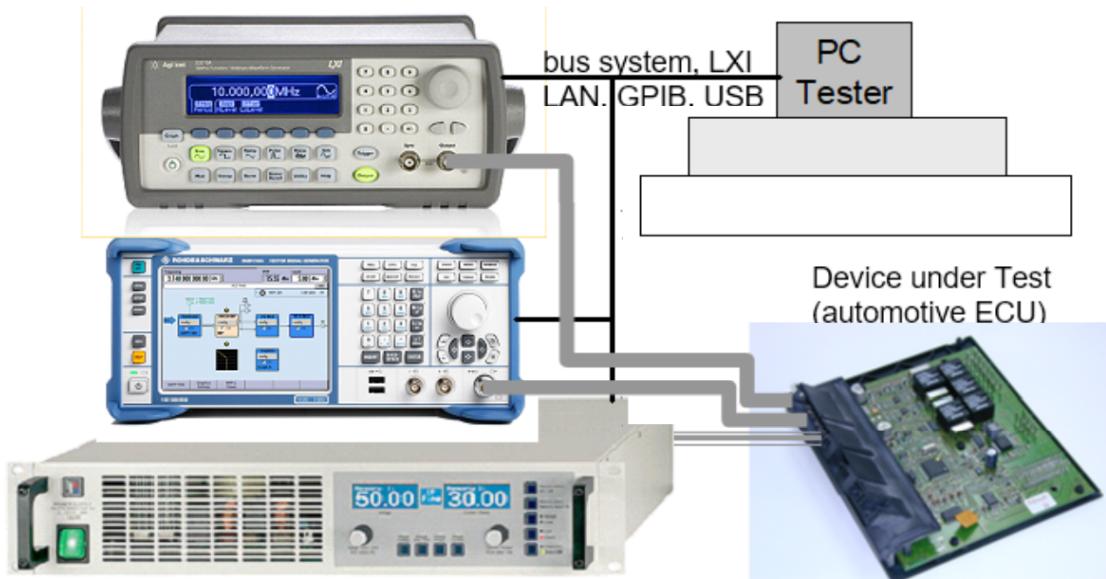
software, but only the behavior of the software at specified actions E.g.: the ECU with anti-squeeze software is tested with a door

1. environmental tests with the complete temperature range (-40° to 95°)
2. various noise on supply conditions (like pulses)
3. different test objects

Test Automation

Particularly, tests that need to be repeated often, test automation is advisable, if a test to be done manually is not feasible, difficult, or too expensive (e.g. lifetime tests). Test automation is also especially useful for regression testing.

1. Regression Testing is used after software changes to assure already achieved functionality.



PC-Software:

1. LabWindows, LabView, C, C#, VisualBasic
2. MATLAB + Data Acquisition Toolbox
3. DSpace
4. VISA (Python + VISA, OpenSource)