

## ICs for Communications

Analog Line Interface Solution ALIS

PSB 4595 Version 2.0

PSB 4596 Version 2.1

User's Manual 04.97 Preliminary Data

T4595-XV20-P1-7600

<b>ALIS</b>		
<b>Revision History:</b>		<b>Current Version: 04.97</b>
Previous Version:		
Page (in previous Version)	Page (in new Version)	Subjects (major changes since last revision)

#### **Edition 04.97**

This edition was realized using the software system FrameMaker®.

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**Overview****1 Overview**

The two chip solution PSB 4595 and PSB 4596 forms the complete frontend of a modem or fax machine. This Analog Line Interface Solution (ALIS) consists of DAA, codec and hybrid and bridges the gap between phone line and data pump. The analog PSB 4595 is fabricated in low-power 0.8  $\mu\text{m}$  BiCMOS technology, the digital PSB 4596 in 0.8  $\mu\text{m}$  CMOS technology. The ALIS concept is a fully programmable modem frontend which allows one single design for the worldwide market:

- Adaptation to specific countries and applications is achieved by downloading appropriate coefficient sets.
- Isolation is achieved by a digital capacitive interface, without a transformer; this makes the ALIS especially suitable for PCMCIA modem design.
- Using an advanced digital filter concept in combination with the programmable electronic DAA the ALIS provides both: excellent transmission performance and high adaptability. This second-generation digital filter concept also allows a maximum of independence between the different filter blocks. This performance makes ALIS suitable for V.34+ and V.PCM modem applications.

A minimum number of external components is required to complete the functionality of the ALIS. The internal precision is based on a very accurate bandgap reference. The frequency behaviour is determined mostly by digital filters which exhibit no fluctuations. As a result of the ADC and DAC concepts linearity is limited only by second-order parasitic effects.

The ALIS chipset can be easily adapted and connected to various modem data pumps or to host based modem solutions. The flexible digital interface of ALIS allows easy programming via the modem data pump or a controller.

Siemens offers a range of reference and evaluation tools for the ALIS chipset. For appropriate tools please contact your next Siemens representative.

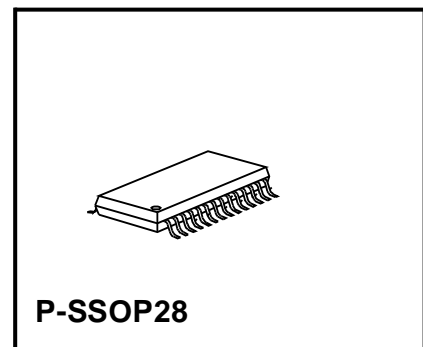
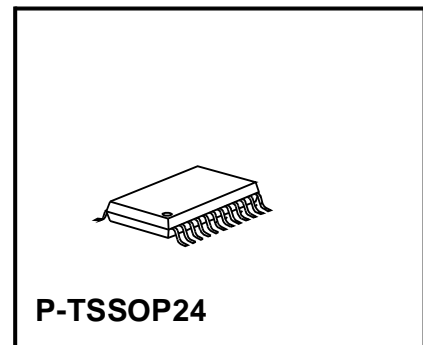
## Analog Line Interface Solution ALIS

PSB 4595  
PSB 4596

CMOS

### 1.1 Features

- ALIS substitutes Data Access Arrangement (DAA), Codec and hybrid
- Ring Detection: Level, Frequency and Cadence
- Caller ID: Detection, Decoding and Storage
- Programmable to the different Country Requirements
- Programmable DC characteristics
- ALIS supports V.34+ and V.PCM
- ALIS meets ETS 300 001 and FCC requirements
- Isolation by digital capacitive interface
- Analog part supplied from the Tip/Ring line by integrated voltage regulator
- High performance Analog to Digital and Digital to Analog Conversion
- DSP based solution to adapt the transmission behaviour especially for
  - AC impedance matching
  - transhybrid balancing
  - frequency response
  - gain
- Advanced test capabilities:
  - digital loops
  - analog loops
- High pass in receive to suppress disturbances from the mains (50/60 Hz)
- Isolated control pins for general purpose use



- Advanced low power 0.8μm analog BICMOS technology for ALIS analog and 0.8μm CMOS technology for ALIS digital
- Two Chip solution: P-TSSOP24 and P-SSOP28 package are PCMCIA conform

Type	Ordering Code	Package
PSB 4595 V2.0	Q67106-H6754	P-TSSOP24
PSB 4596 V2.1	Q67106-H6755	P-SSOP28



## 1.2 Logic Symbol

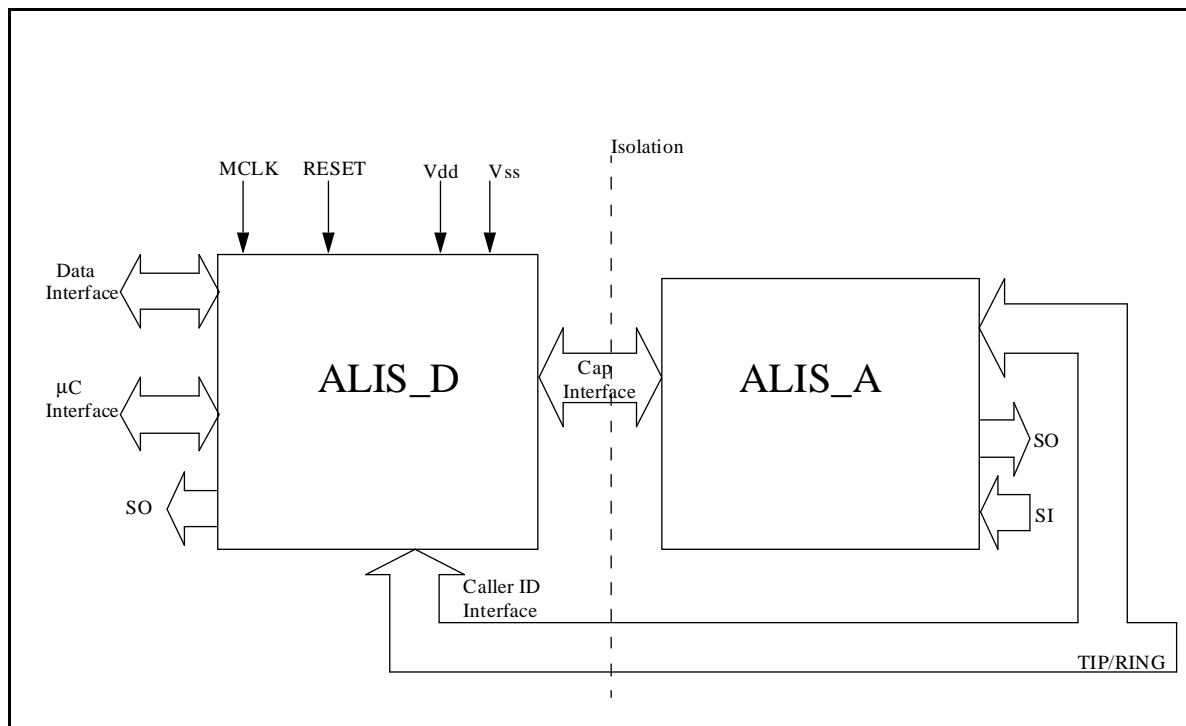


Figure 1: Logic Symbol of the ALIS Chipset

## 1.3 Pin Configuration

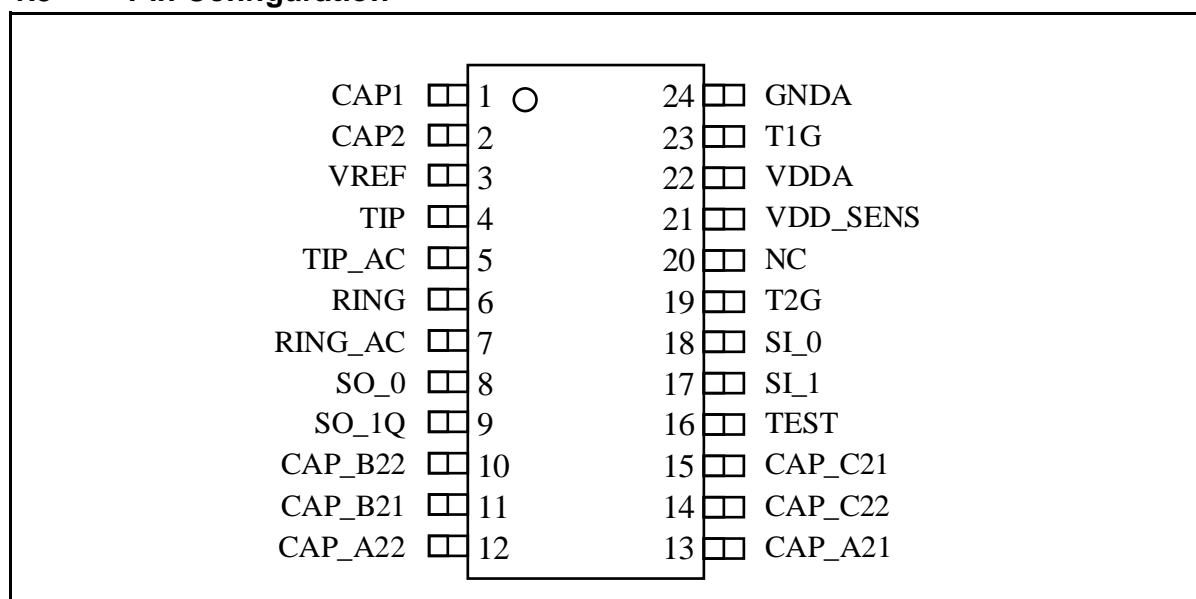


Figure 2: Pin Configuration of ALIS\_A (top view)

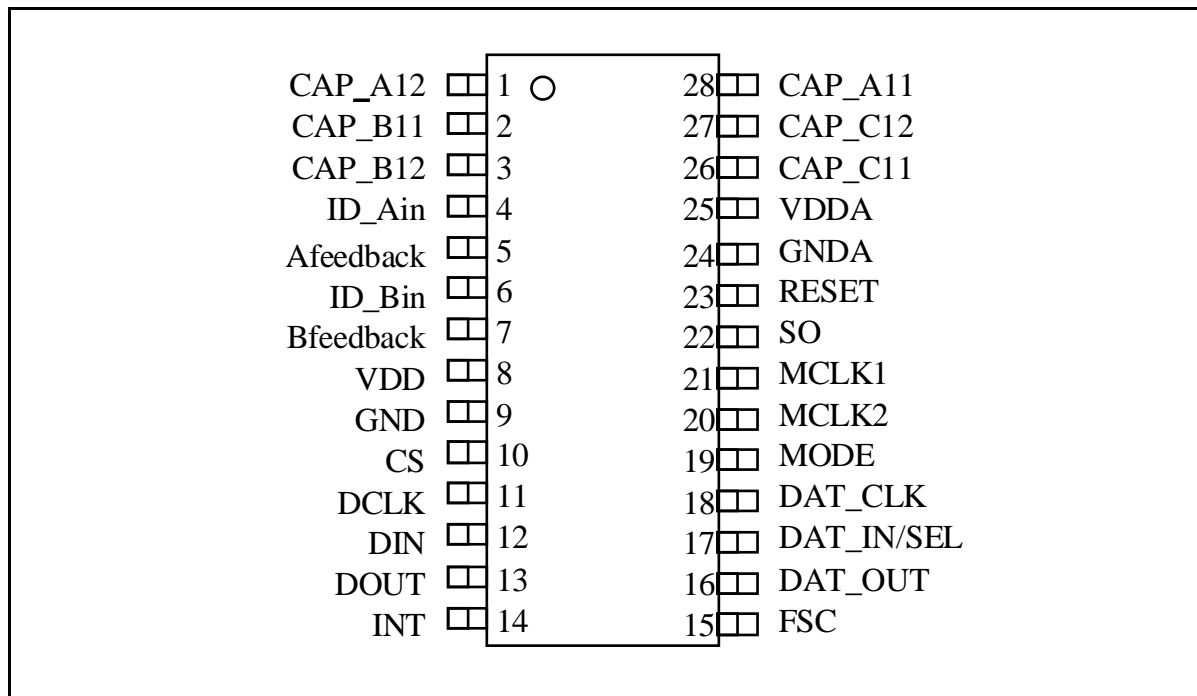


Figure 3: Pin Configuration of ALIS \_D (top view)

## Pin Definition and Functions

### 2 Pin Definition and Functions

#### 2.1 ALIS\_A PSB 4595

Pin No.	Symbol	Function	Descriptions
22	VDDA	Power	Programmable supply for the circuitry
24	GNDA	Power	Analog ground: All signals are referred to this pin
4	TIP	I	TIP ac+dc sense input
5	TIP_AC	I	TIP ac sense input
6	RING	I	RING ac+dc sense input
7	RING_AC	I	RING ac sense input
23	T1G	O	Gate for external transistor T1 (ac/dc control)
19	T2G	O	Gate for external transistor T2 (VDDA control)
21	VDD_SENS	I	VDDA sense input
3	VREF	I/O	Reference voltage: Must be connected to an external Capacitance with more than 10 nF (typ. 15 nF) to GNDA
1	CAP1	I/O	Pin for external capacitor with more than 1 $\mu$ F for DC filtering to pin Cap2
2	CAP2	I/O	See Cap1
18	SI_0	I	Auxiliary input pin 0
17	SI_1	I	Auxiliary input pin 1
8	SO_0	O	Auxiliary output pin 0
9	SO_1Q	O	Auxiliary output pin 1
16	TEST	I	Must be connected permanently to GNDA
13	CAP_A21	I	Must be connected by a capacitance with more than 5pF to CAP_A11.
12	CAP_A22	I	Must be connected by a capacitance with more than 5pF to CAP_A12.
11	CAP_B21	O	Must be connected by a capacitance with more than 5pF to CAP_B11.
10	CAP_B22	O	Must be connected by a capacitance with more than 5pF to CAP_B12.

---

**Pin Definition and Functions**

<b>Pin No.</b>	<b>Symbol</b>	<b>Function</b>	<b>Descriptions</b>
15	CAP_C21	I	Must be connected by a capacitance with more than 5pF to CAP_C11.
14	CAP_C22	I	Must be connected by a capacitance with more than 5pF to CAP_C12.

Table 1: ALIS\_A pin definition

## Pin Definition and Functions

### 2.2 ALIS\_D PSB 4596

Pin No.	Symbol	Function	Description
8	VDD	Power	+5 Volt supply for the digital circuitry
9	GND	Power	Ground digital: All signals are referred to this pin
25	VDDA	Power	+5 Volt supply for the analog circuitry
24	GNDA	Power	Ground analog: All analog signals are referred to this pin
21	MCLK1	I	Master clock1: One pin of a crystal or ceramic resonator is connected. This pin can also be driven from an external clocking source of 16.384 MHz, synchronous to FSC
20	MCLK2	O	Master clock2: Other pin of a crystal or ceramic resonator is connected. When MCLK1 is driven by an external clock this pin should be left open
23	RESET	I	Reset input: Forces the device to the default mode (low active)
15	FSC	BI	As input: Frame synchronisation clock, 8 kHz, identifies the beginning of the frame, FSC must be synchronous to DATCLK As Output: Indicates the beginning of a new frame
17	DAT_IN / SEL	I	Data-interface: Receive Data from the DSP, the data is received in 16 bit bursts every 125 $\mu$ s. Interface selection pin in mux mode.
16	DAT_OUT	O	Data-interface: Transmit Data to the DSP, the data is transmitted in 16 bit bursts every 125 $\mu$ s
18	DAT_CLK	I	Data clock 128 to 1024 kHz: Determines the rate at which Data is shifted into or out of the Data Interface
10	CS	I	$\mu$ -controller interface: Chip select enable to read or write data, active low
11	DCLK	I	$\mu$ -controller interface: Clock, maximum clock rate 1024 kHz

## Pin Definition and Functions

Pin No.	Symbol	Function	Description
12	DIN	I	μ-controller interface: Input data
13	DOUT	TRI	μ-controller interface: DOUT is high 'Z' if no data is transmitted
14	INT	O	μ-controller interface: Interrupt output pin
19	MODE	I	Interface Mode pin (parallel or mux mode)
4	ID_Ain	I	Input for Caller ID comparator (connection to TIP)
6	ID_Bin	I	Input for Caller ID comparator (connection to RING)
5	Afeedback	O	Feedback for Caller ID comparator
7	Bfeedback	O	Feedback for Caller ID comparator
28	CAP_A11	O	Must be connected by a capacitance with more than 5pF to CAP_A21. <sup>1)</sup>
1	CAP_A12	O	Must be connected by a capacitance with more than 5pF to CAP_A22.
2	CAP_B11	I	Must be connected by a capacitance with more than 5pF to CAP_B21.
3	CAP_B12	I	Must be connected by a capacitance with more than 5pF to CAP_B22.
26	CAP_C11	O	Must be connected by a capacitance with more than 5pF to CAP_C21.
27	CAP_C12	O	Must be connected by a capacitance with more than 5pF to CAP_C22.
22	SO	O	Auxiliary output pin <sup>2)</sup>

1) For a detailed description of the cap interface see "ALIS Cap interface" on page 83

2) can be used to control the hook switch

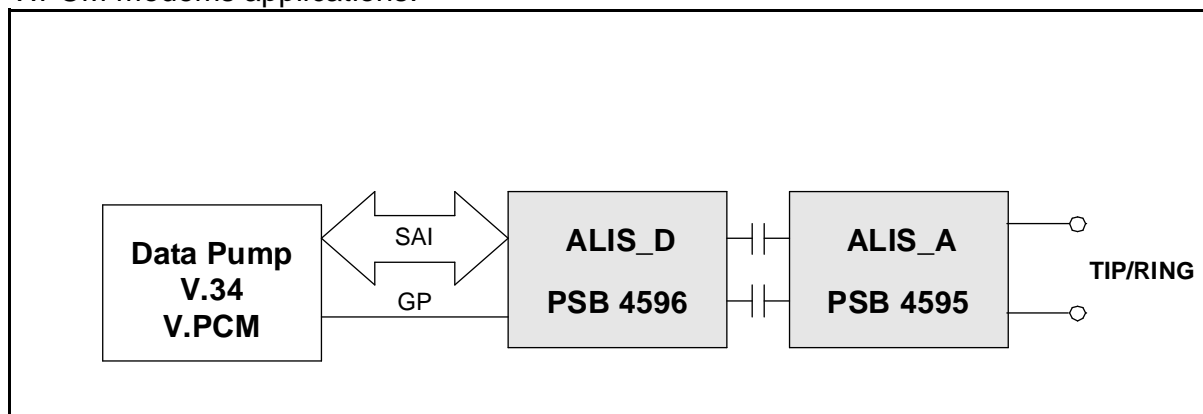
Table 2: ALIS\_D pin definition

## 3 System Integration

ALIS can be used in different modem applications to connect the data pump to the TIP/RING wire.

### 3.1 ALIS with V.34 Chipset

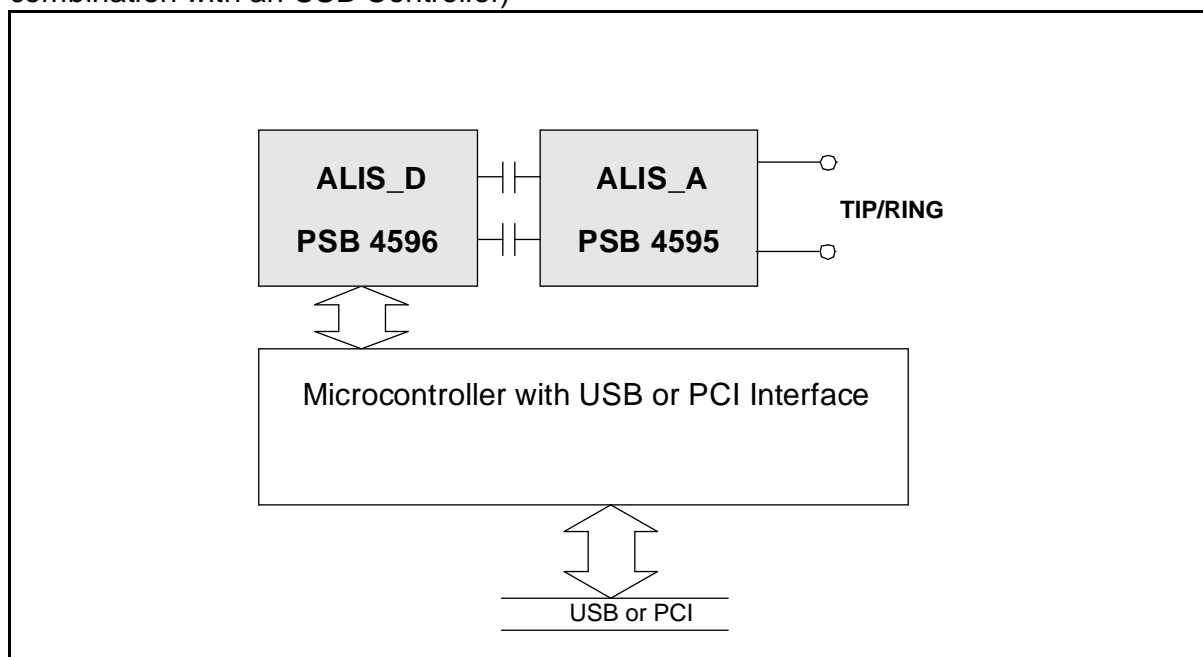
In combination with a V.34 chipset ALIS is an optimal solution for standard V.34 and V.PCM modems applications.



*Note: SAI Serial Audio Interface  
GP General Purpose I/O pin's*

### 3.2 ALIS with Software Modem

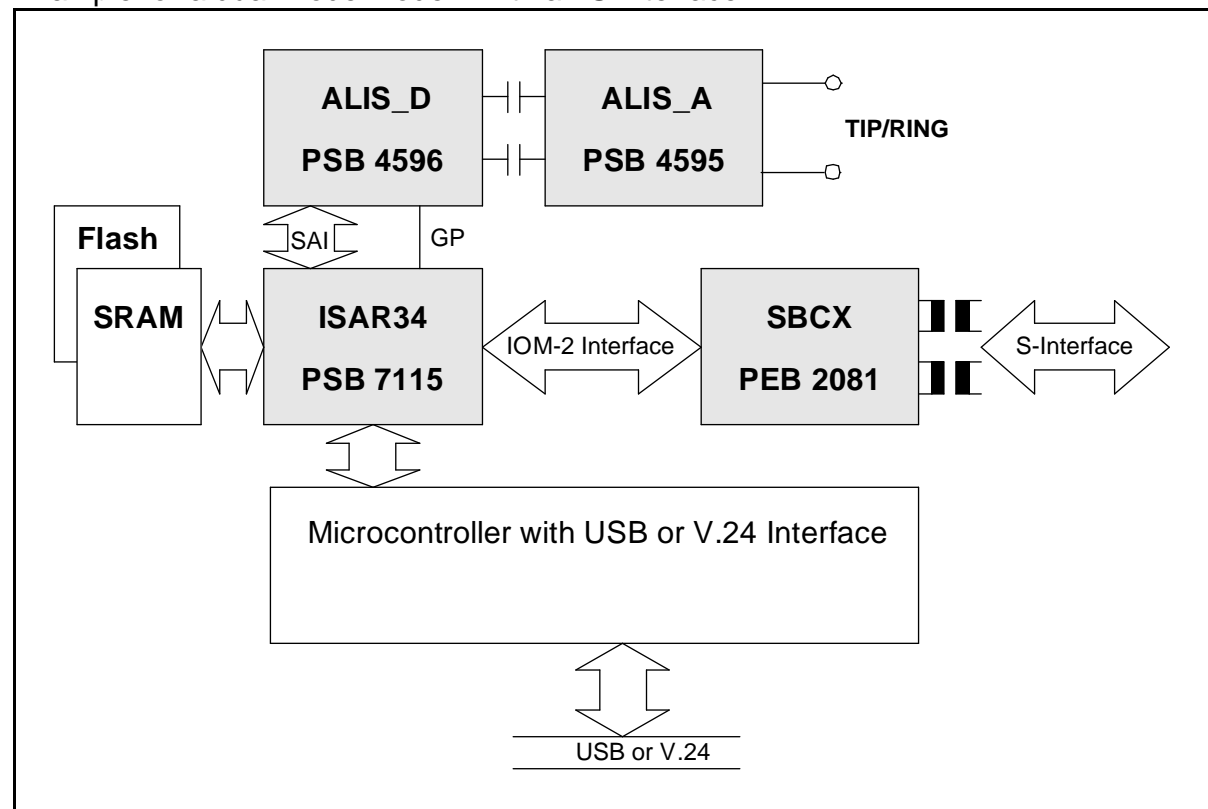
ALIS also supports software modems where V.34 runs on the host computer (e.g. in combination with an USB Controller)



### 3.3 Dual Mode Modem

In combination with the SIEMENS ISDN chipset ALIS supports dual mode modems to connect either to the TIP/RING line or to a S or a U-interface for ISDN applications.

Example for a dual mode modem with an S-interface

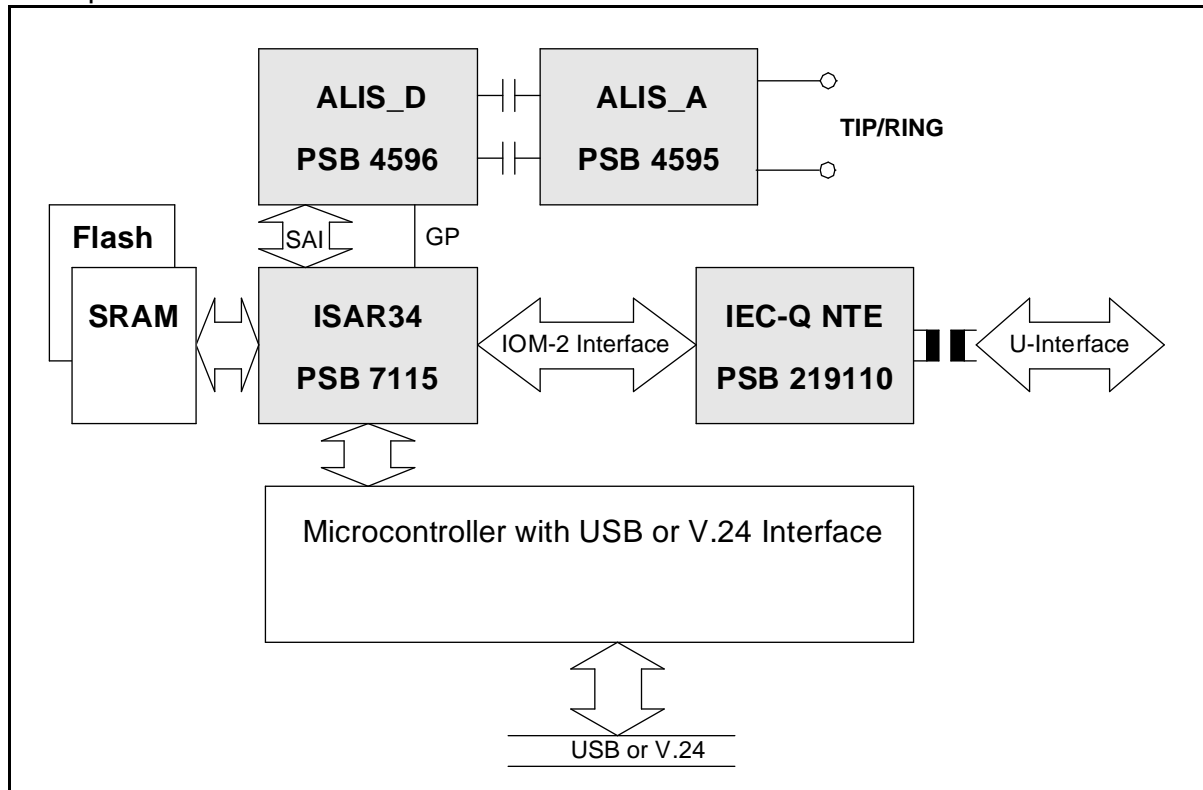


ISAR34 Enhanced data access controller (PSB 7115)

SBCX S/T Bus Interface Circuit Extended (PEB 2081)



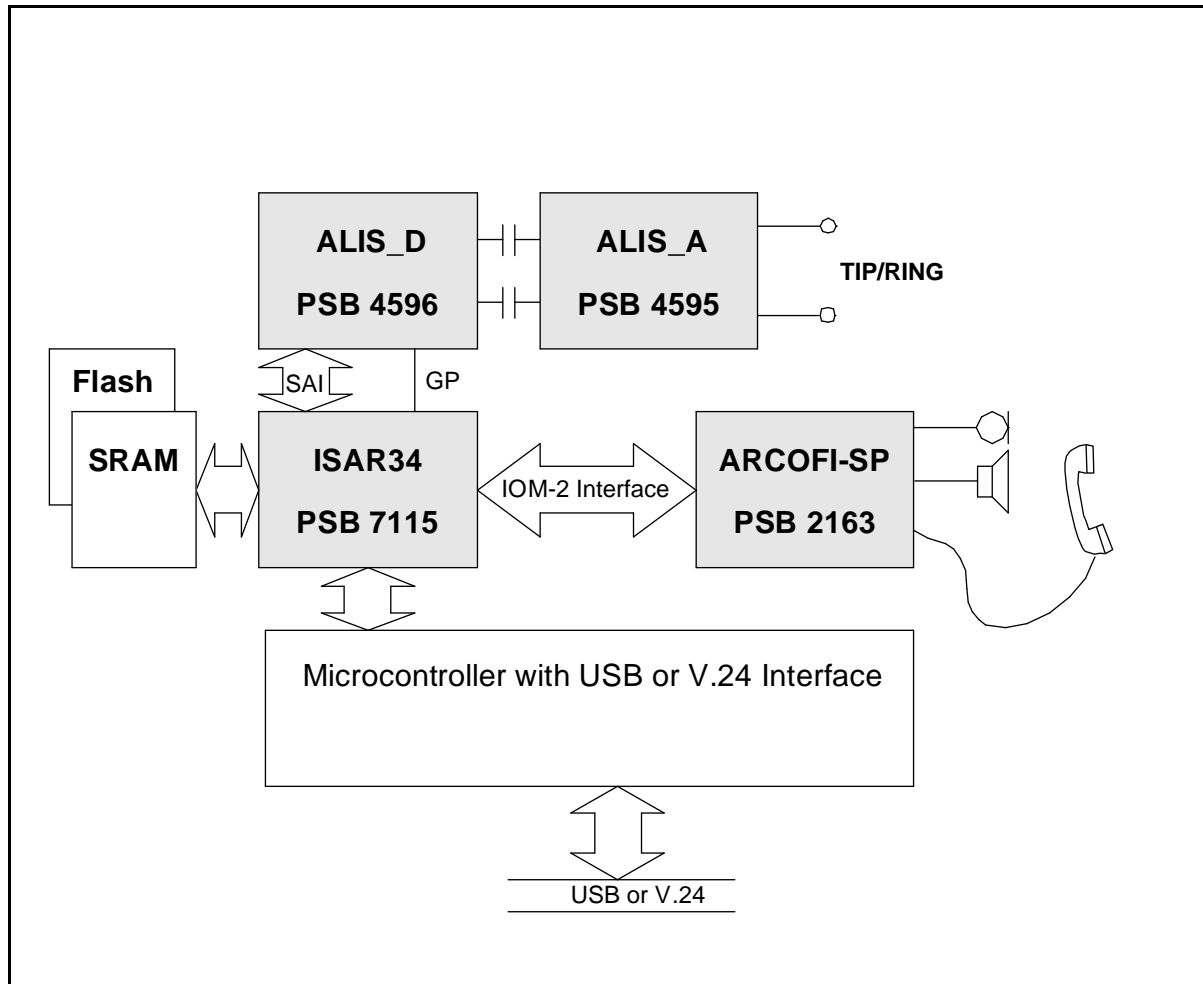
Example for a dual mode modem with an U-interface



ISAR34 Enhanced data access controller (PSB 7115)

IEC-Q NTE ISDN Echo Cancellation Circuit for Network Termination and Terminals

## 3.4 Modem with Speakerphone

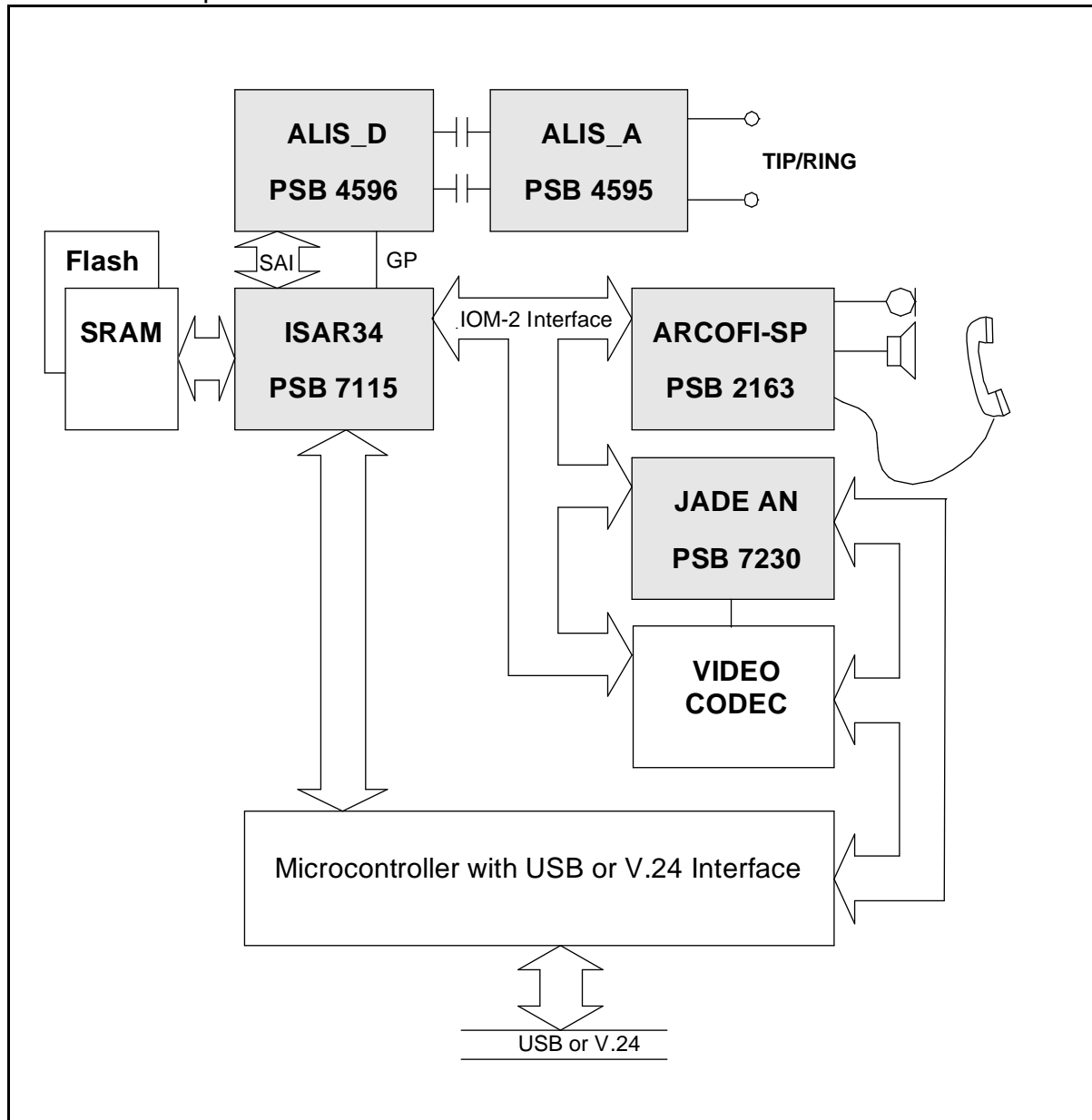


ARCOFI-(SP) Audio Ringing Codec (PSB 2160, PSB 2163, PSB 2165, PSB 2168)

ISAR34 Enhanced data access controller (PSB 7115)

## 3.5 Analog Videophone

The picture below shows a system solution for an analog videophone application using a SIEMENS chipset.



ARCOFI-(SP) Audio Ringing Codec (PSB 2160, PSB 2163, PSB 2165, PSB 2168)

ISAR34 Enhanced data access controller (PSB 7115)

JADE Joint Audio Decoder Encoder (PSB 7230, PSB 7238)

## 4 ALIS Realization

The ALIS chip set replaces all the major parts of traditional frontend for modem solutions. The circuit consists of two major parts, a DSP based Codec and an electronic DAA. Advanced features like ring detection, pulse dialing and caller-ID are integrated on-chip. To minimize power consumption additional operating modes like Sleep Mode or Ringing Mode are implemented.

### 4.1 ALIS Block Diagram

The analog front end (ALIS\_A) is connected to the line via TIP/RING. The programmable supply voltage for ALIS\_A is generated from the line by the Vdd-Control. Two/four wire transformation is realized in the Hybrid. Analog antialiasing Prefilters (PREFI) and smoothing Postfilters (POFI) are included for signal conditioning. High performance oversampling Analog-to-Digital Converters (ADC) and Digital-to-Analog Converters (DAC) assure the required conversion accuracy. The ADC and DAC are connected to the Digital Signal Processor (DSP) on the digital part (ALIS\_D) via a dedicated capacitive interface which also provides the required isolation to the line. Special Hardware Filters perform filtering functions like interpolation and decimation. The DSP handles all the necessary algorithms, e.g., bandpass filtering, sample rate conversion, ring detection, caller ID decoding etc. Also, all programmable filters and functions are controlled and processed by the DSP. The control interface allows external control of the ALIS features and provides transparent access to ALIS commands and signalling pins. Thus precalculated sets of coefficients can be downloaded from the system to the on-chip Coefficient RAM (CRAM) to program the filters. Transmit and receive data is transferred from and to the data pump via the data interface.

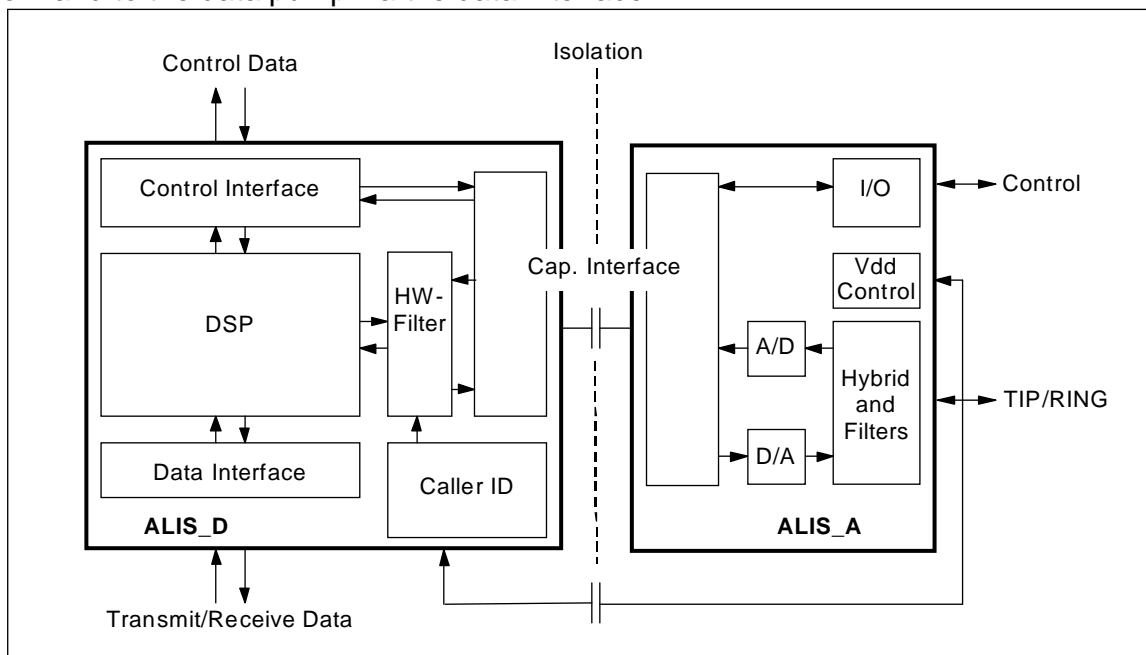


Figure 4: ALIS Block Diagram

## 4.2 ALIS AC Signal Flow Graph

This second-generation digital filter concept also allows a maximum of independence between the different filter blocks. Each filter block has a one-to-one correspondence with a particular network element.

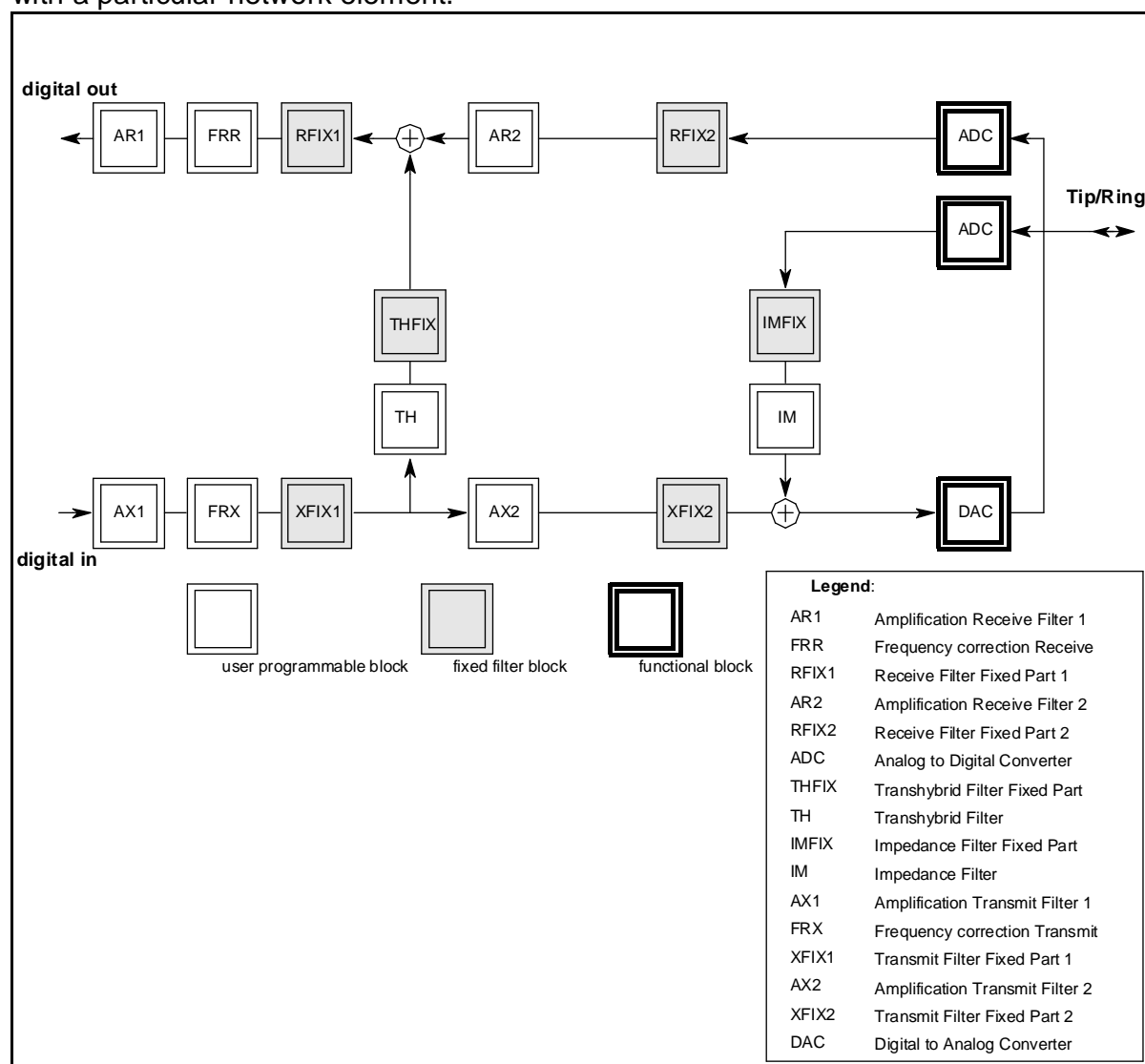


Figure 5: AC Signal flow graph

Definition:

Transmit - path: D/A path

Receive - path: A/D path

### 4.2.1 Receive Path

After passing the DAA and a simple antialiasing Prefilter with an analog gain stage the voice signal is converted to a 1-bit digital data stream in the Sigma-Delta-converter. The

first downsampling steps are performed in fast running digital hardware filters. The following steps are implemented in micro-code which is executed by the central DSP. This allows an easy and flexible programming of many parameters. At the end the fully processed signal is transferred to the data interface.

#### **4.2.2 Transmit Path**

The digital input signal is received via the data interface. Low-pass filtering, gain correction and frequency response correction are implemented in micro-code and executed by the DSP. The upsampling interpolation is then performed by fast hardware structures to reduce the DSP load. The upsampled 1-bit data stream is converted to an analog equivalent which is smoothed by a Postfilter (POFI) and converted to a 2 wire signal in the DAA.

#### **4.2.3 Loops**

The ALIS implementation includes two loops. One is used to generate the AC-termination impedance (IM) and the other is used to perform proper hybrid balancing (TH). A simple additional path IM (from the receive to the transmit path) supports the impedance matching function.

#### **4.2.4 Test Features**

There are several analog and digital test loops implemented in ALIS. For test purposes it is possible to short-cut the receive and the transmit path at two different points.

## 4.3 ALIS Ring and CID Signal Flow Graph

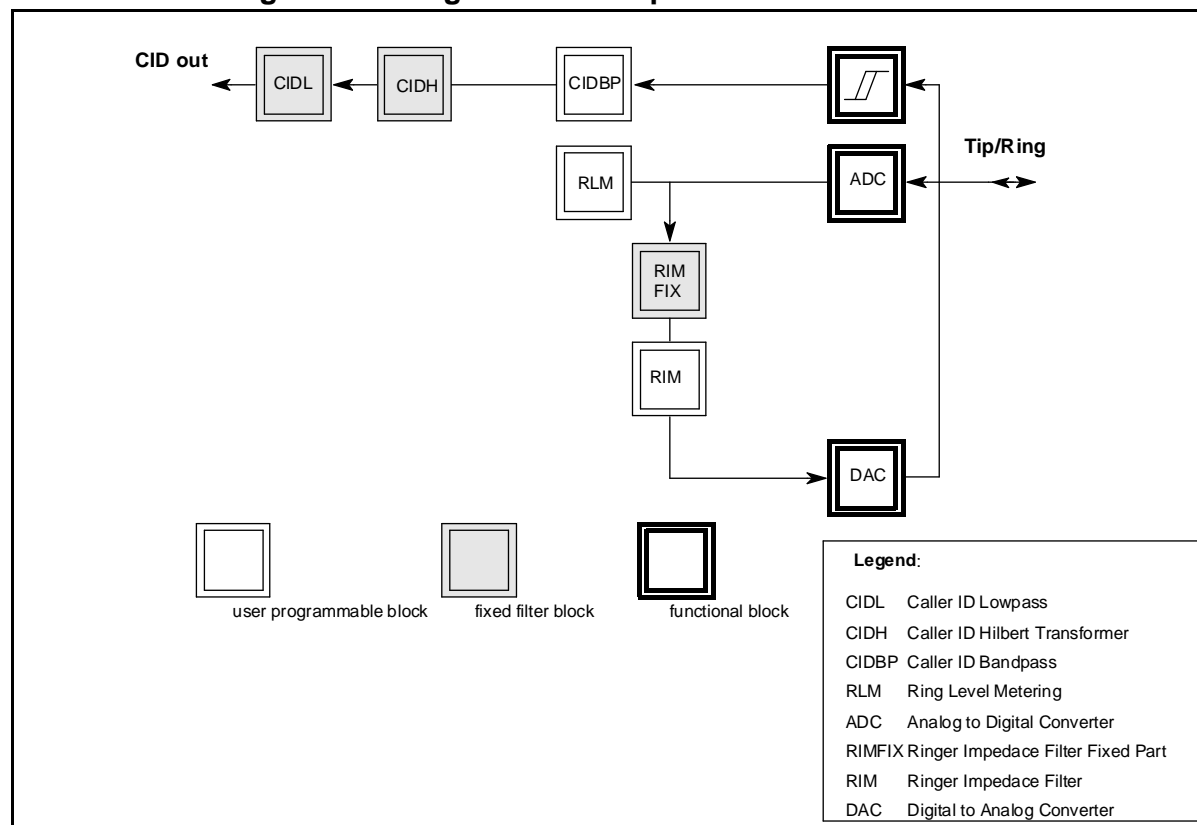


Figure 6: Ring Signal flow graph

### 4.3.1 Caller ID (CID) Path

The FSK-Signal which contains the caller information is converted to a 1-bit data stream by a comparator to minimize power consumption. Downsampling steps are performed in fast running digital hardware filters. To decode the caller ID band-pass filtering, hilbert-transforming and other functions are executed by the DSP.

### 4.3.2 Ring Level-Metering (RLM) Path

Ring detection is realized in this path. The digital input is band-pass filtered, integrated and compared to a threshold to determine if a ring signal occurred. Threshold and band-pass filter are programmable. The result of this operation can be monitored by reading the RMR bit. See "CR1 Configuration Register 1 (dialing)" on page 38

### 4.3.3 Loops

There is one loop to generate the Ring-termination Impedance (RIM).

### 5 Configurations Overview

#### 5.1 Connection to the telephone line

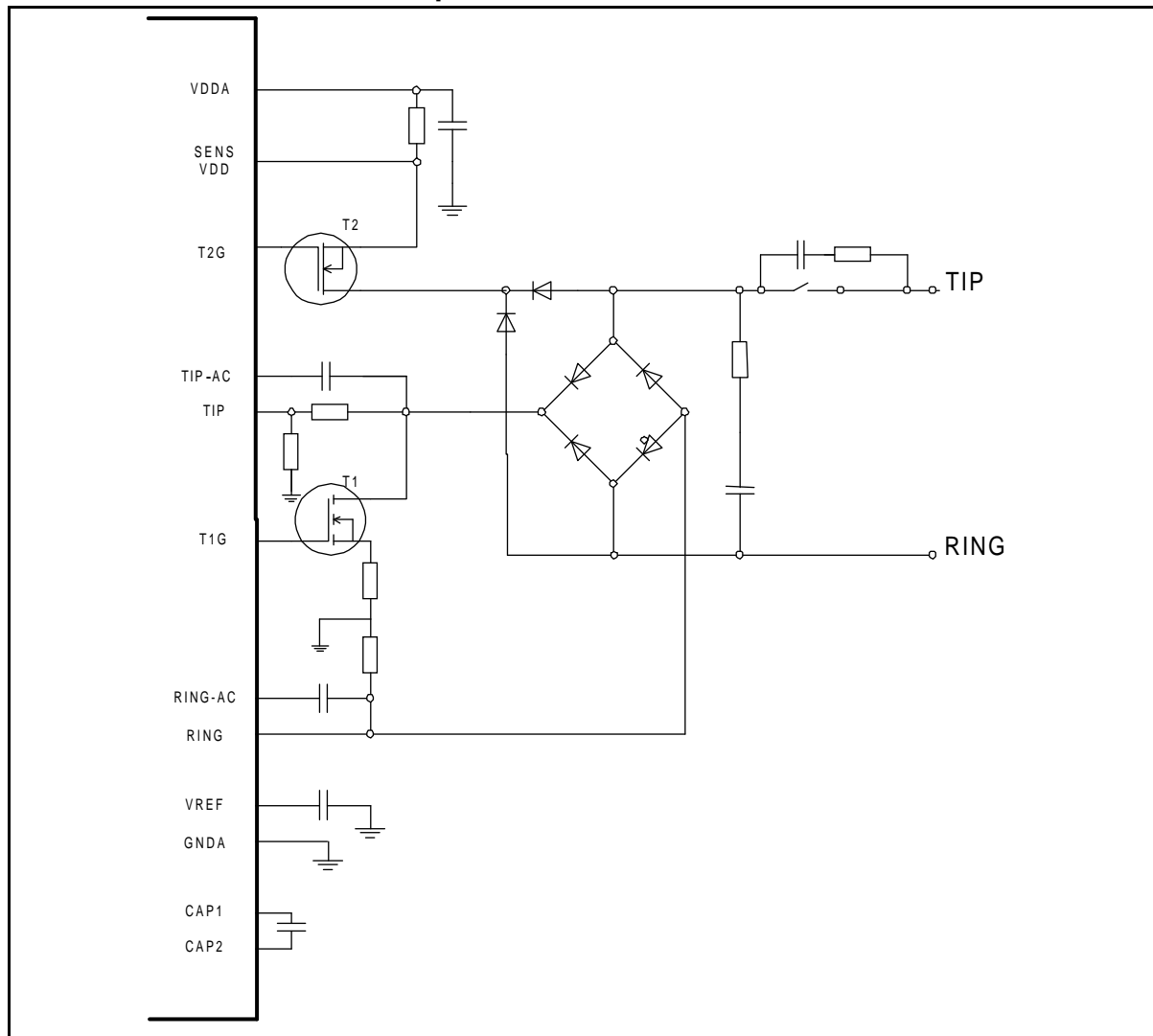


Figure 7: Connection to the telephone line

Recommended Transistors:

T1: BSP 88

T2: BSP 129

In addition to the protection circuit and the bridge, two transistors (T1, T2) to handle the line currents, a few resistors for current measurement and some capacitors for AC coupling and VDD buffering are necessary. For start-up reasons the transistor T2 has to



be of depletion type. External components for EMC protection are not shown, as they strongly depend on the board layout.

## **5.2 Host Interface**

The host interface consists of a serial  $\mu$ -controller interface and a 16-bit linear data interface. They are used to connect the ALIS either to a  $\mu$ -controller and or to a data pump. The two serial interfaces can be accessed on two parallel ports or in time-multiplex (MUX) mode on one single port.

### **5.2.1 The $\mu$ -Controller Interface**

The internal configuration registers, the auxiliary ports, and the Coefficient RAM (CRAM) of the ALIS are programmable via the serial  $\mu$ -controller interface. The  $\mu$ -controller interface consists of 4 pins:

CS:	Chip select, for enabling interface (active low)
DCLK:	Clock, 1 kHz to 1024 kHz
DIN:	Data input
DOUT:	Data output

CS is used to start a serial access to the ALIS registers and the Coefficient RAM. Following a falling edge of CS the first eight bits received on DIN specify the command. Subsequent data bytes (number depends on command) are stored in the selected configuration registers or the selected part of the coefficient RAM.

Serial interface specification: 8 bit, no parity, no start/stop bit. Every command must begin with a falling edge of CS.

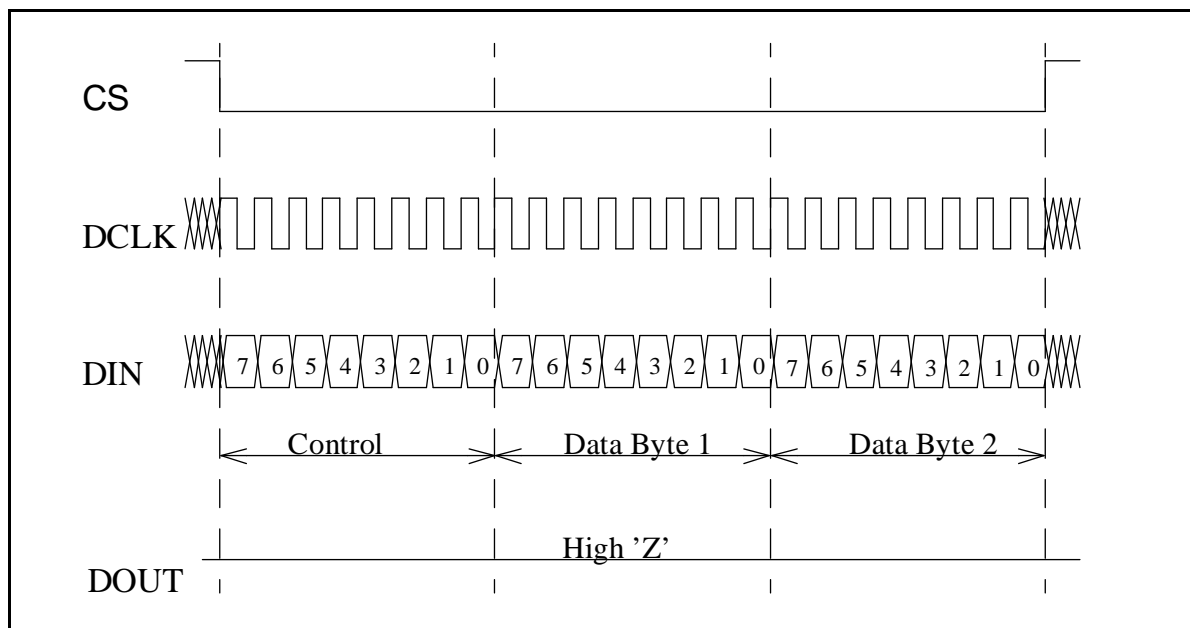


Figure 8: Example for a write access, with two data bytes transferred

If the first eight bits received via DIN specify a read-command, the ALIS will start to response via DOUT with its specific identification byte. The number of specified data bytes within the command (contents of configuration registers, or contents of the CRAM) will follow on DOUT.

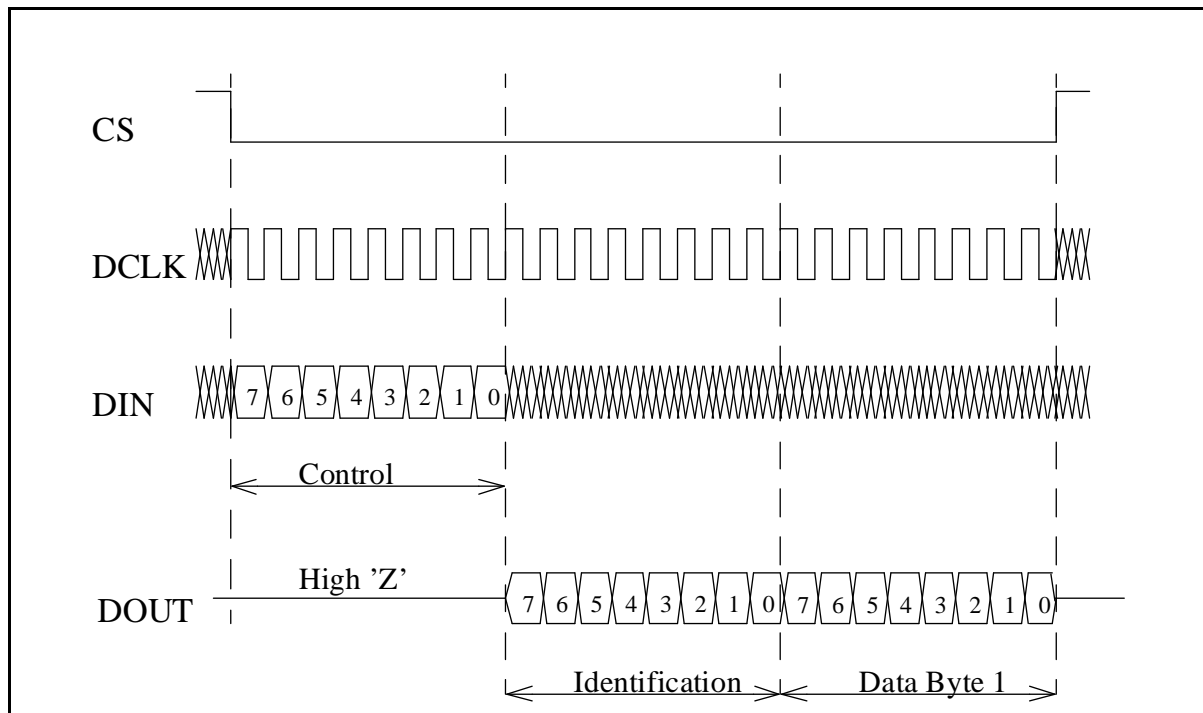


Figure 9: Example for a read access, with one data byte transferred via DOUT

## Configurations Overview

The data transfer is synchronized by DCLK. DIN is latched at the falling edge of DCLK, while DOUT changes with the rising edge of DCLK. During execution of a command which is followed by output data (read command), the device will not accept any new command via DIN. The data transfer sequence can be interrupted by setting CS to high. To reduce the number of connections to the  $\mu$ -Processor DIN and DOUT may be strapped together to a bidirectional datapin.

### 5.2.2 The Data Interface

A serial data-interface is used for the transfer of voice data. The data-interface consists of 5 pins:

DAT\_CLK: Clock, 128 kHz to 1024 kHz  
FSC: Frame Synchronisation Clock, 8 kHz  
DAT\_IN: Transmit Data input  
DAT\_OUT: Receive Data output

The Frame Sync (FSC) pulse identifies the beginning of a receive and a transmit frame. DAT\_CLK synchronizes the data transfer on DAT\_IN and DAT\_OUT. The data bytes are serialized to 16 bit width and MSB first. The rising edge indicates the start of the bit, while the falling edge is used to latch the contents of the received data.

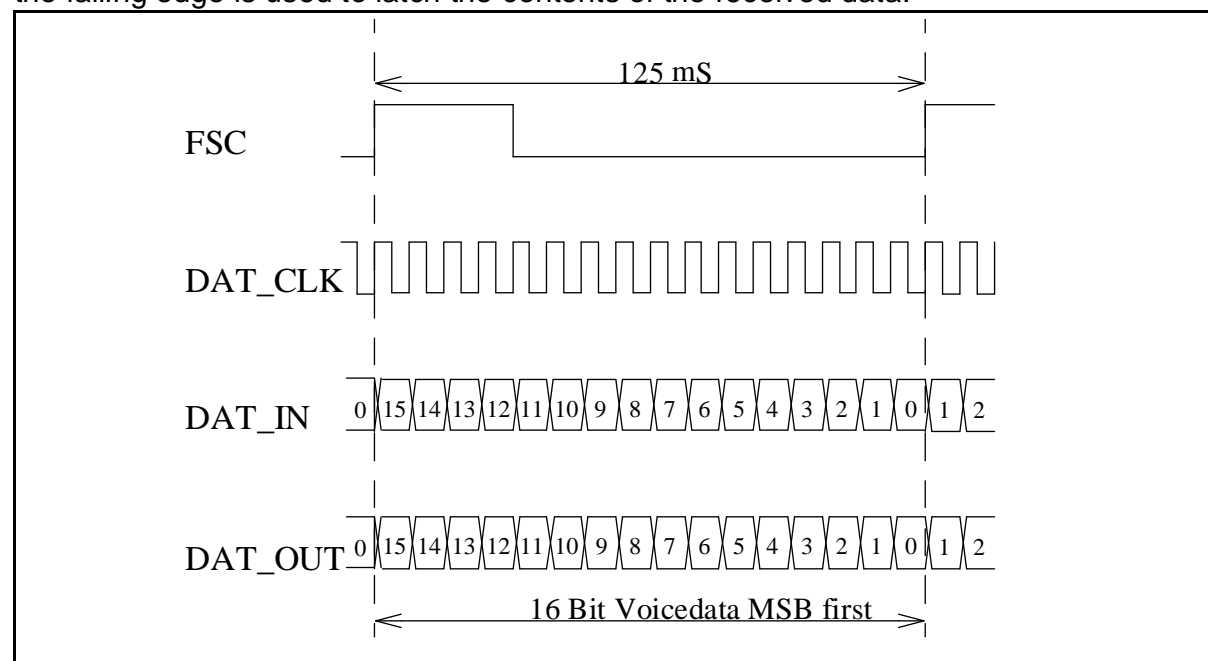


Figure 10: Example for a clock rate of 128 kb/s

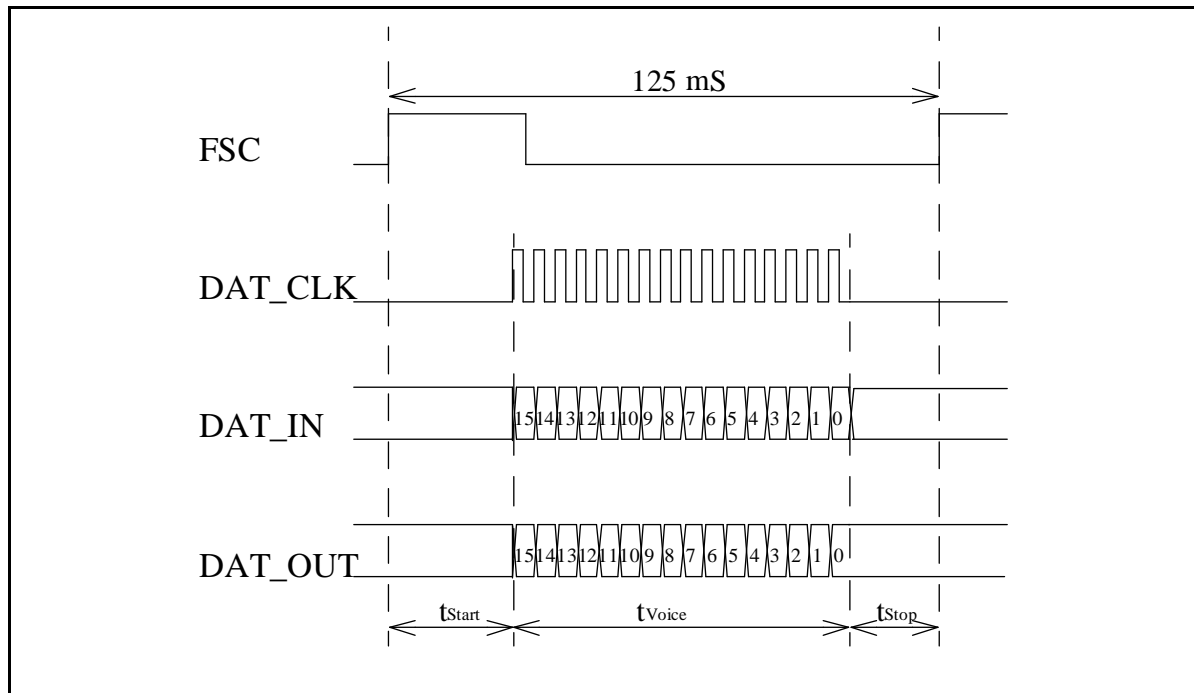


Figure 11: Example for clock rate higher than 128 kb/s  
The data package must stay within the frame,  $t_{Start} > 0$  and  $t_{Stop} > 0$ .  
The FSC signal can be generated externally by the host, or by ALIS.

### 5.2.3 Interface Modes

#### 5.2.3.1 Parallel Mode

By connecting the MODE pin to GND the  $\mu$ C and the data interface are accessed via two parallel ports.

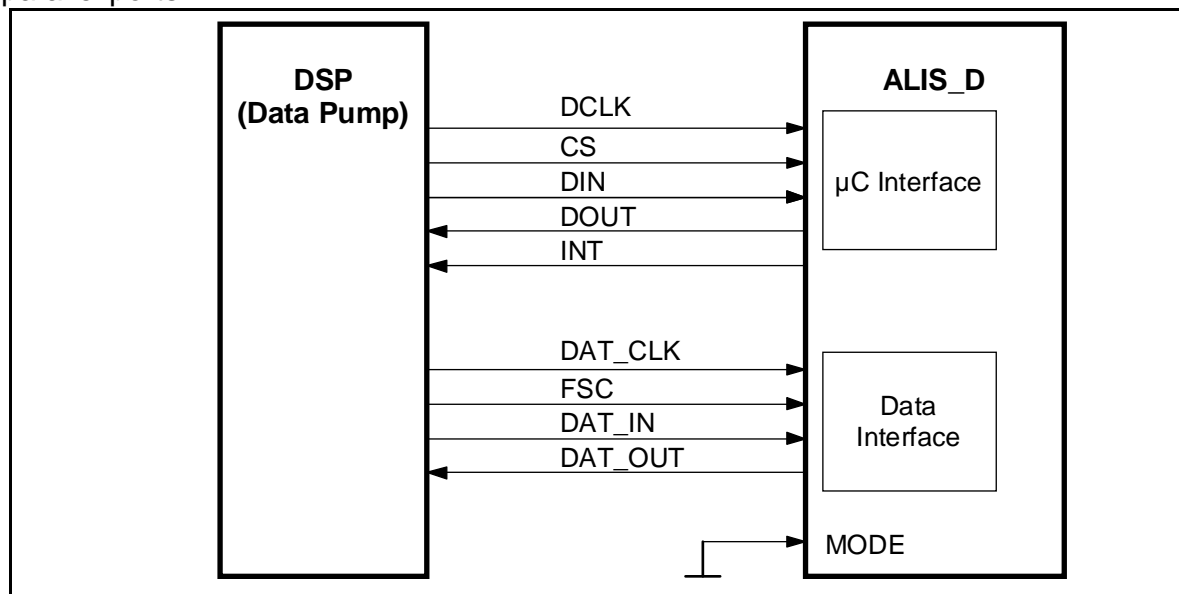


Figure 12: Host interface in parallel mode, FSC as input

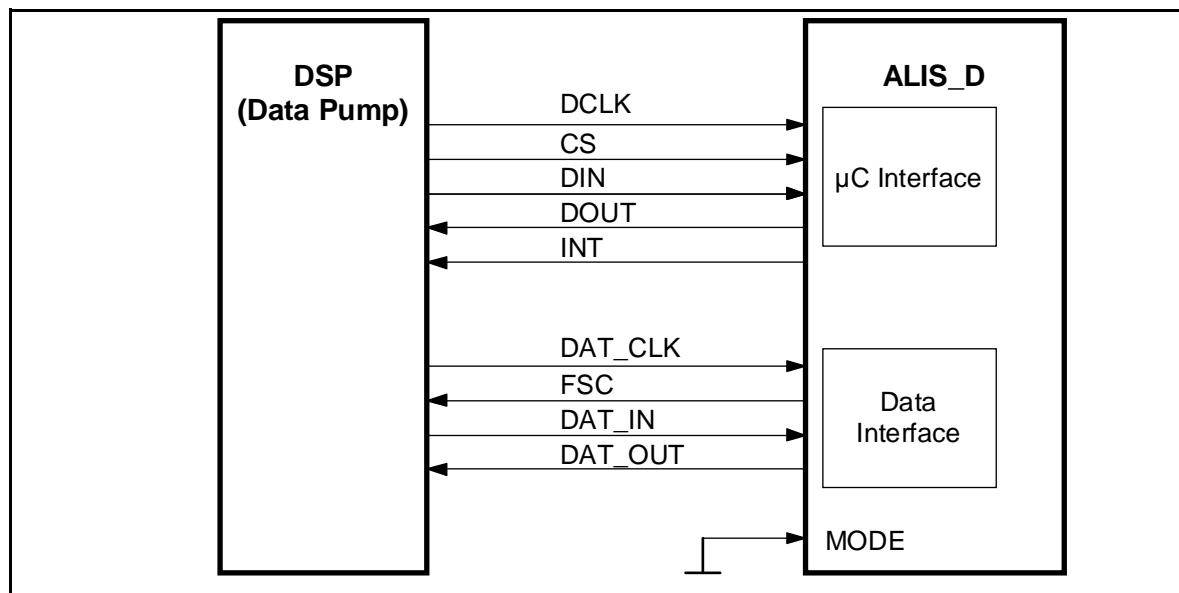


Figure 13: Host interface in parallel mode, FSC as output

### 5.2.3.2 Multiplex Mode

By connecting the MODE pin to VDD the two interfaces are time multiplexed on one single port. The selection between the interfaces is done by the DAT\_IN/SEL pin.

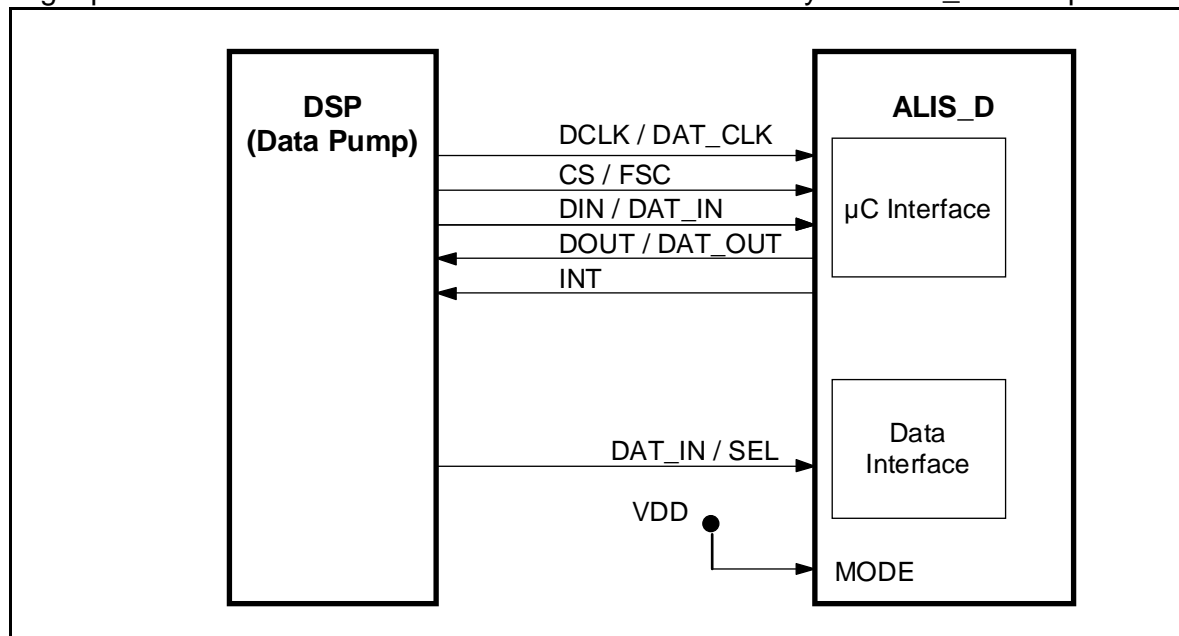


Figure 14: Host interface in MUX mode, FSC as input

DAT_IN / SEL = 0	
PIN No	Function
11	DCLK
10	CS
12	DIN
13	DOUT

DAT_IN / SEL = 1	
PIN No	Function
11	DAT_CLK
10	FSC
12	DAT_IN
13	DAT_OUT

Table 3: Pin definition in MUX mode, FSC as input

## Configurations Overview

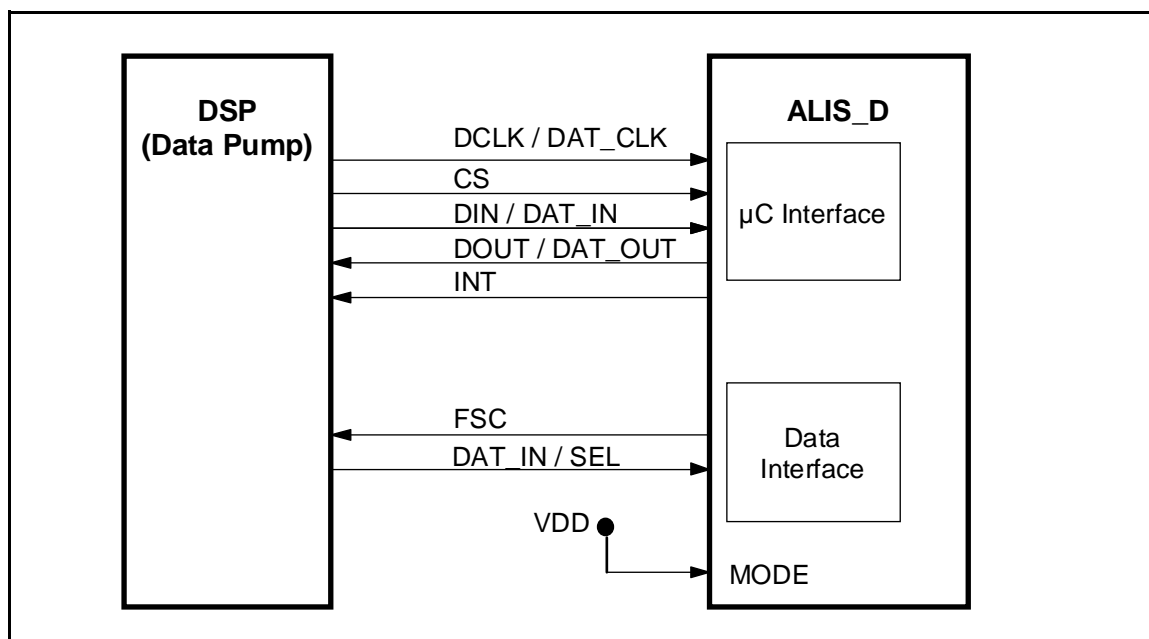


Figure 15: Host interface in MUX mode, FSC as output

DAT_IN / SEL = 0	
PIN No	Function
11	DCLK
10	CS
12	DIN
13	DOUT
15	FSC (output)

DAT_IN / SEL = 1	
PIN No	Function
11	DAT_CLK
10	VDD / GND <sup>1)</sup>
12	DAT_IN
13	DAT_OUT
15	FSC (output)

1) must be connected to a fixed potential

Table 4: Pin definition in MUX mode, FSC as output

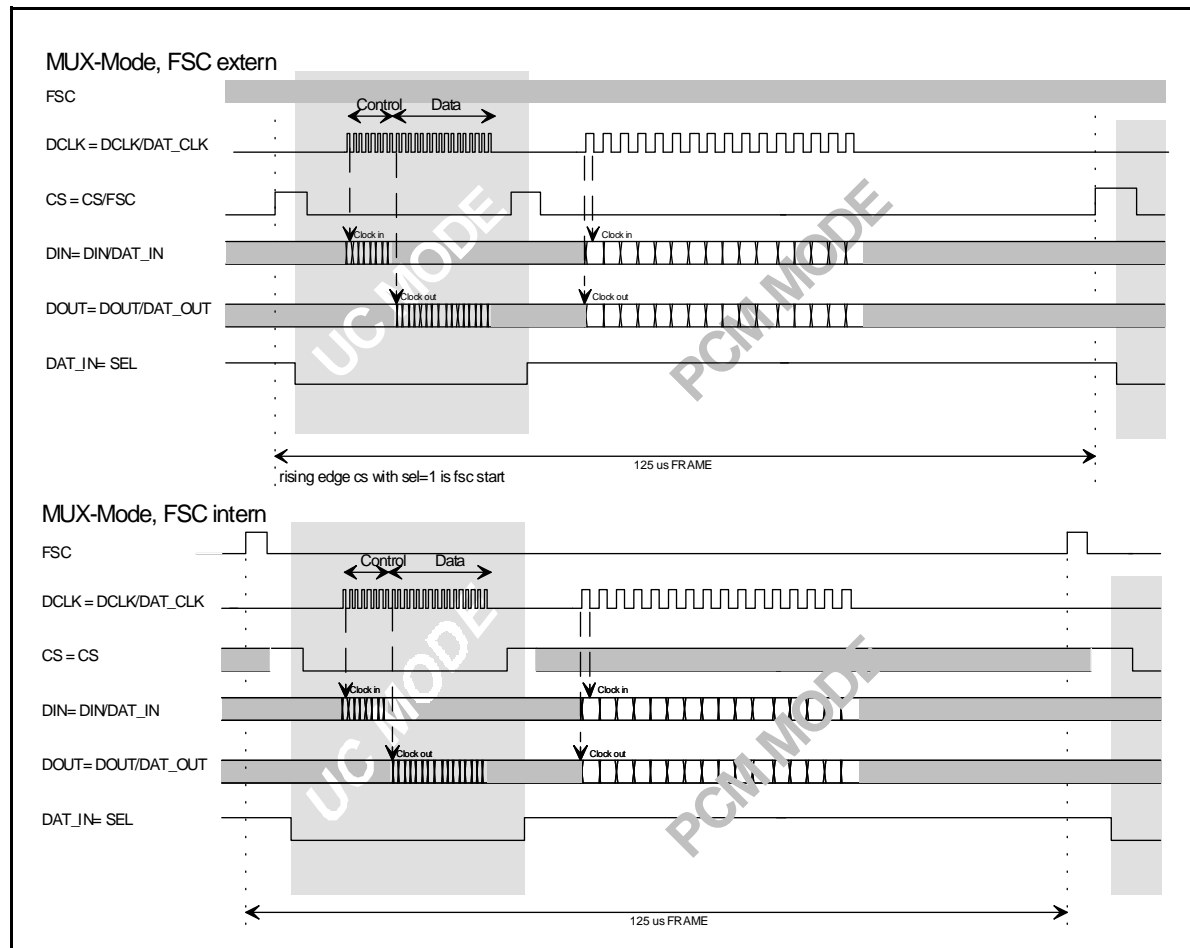


Figure 16: Protocol for transmission of  $\mu$ C- and PCM data in MUX mode

### 5.3 Clocking

ALIS works with a master clock frequency of typically 16.384 MHz. This clock can either be supplied from an external source or generated with a crystal by ALIS\_D.

It is essential that the ratio of the master clock frequency to the FSC frequency is exactly 2048. This is of course guaranteed if the FSC signal is generated internally.

#### 5.3.1 External clock

An external clock signal must be connected to pin MCLK1. The MCLK2 pin must remain unconnected and the CLK\_EXT bit in CR0 must be programmed to a logic '1'. See "CR0 Configuration Register 0 (filters)" on page 37



## 5.3.2 Crystal clock

Because the ALIS includes an on-chip oscillator circuit, an external crystal may be used. The crystal should be connected across the MCLK1 and MCLK2 pins with two capacitors connected as shown in figure 17. The CLK\_EXT bit in CR0 must be programmed to a logic '0' (=default value after reset). Capacitor values are dependent on crystal type and should be specified by the crystal manufacturer. A parallel-resonant, fundamental frequency, microprocessor-grade crystal should be used.

To ensure that the ratio between master clock and FSC signal is correct, ALIS must be programmed to internal FSC generation (set Fsc\_en bit in CR4 to a logic '1'). See "Host interface in parallel mode, FSC as output" on page 29 and "Host interface in MUX mode, FSC as output" on page 31.

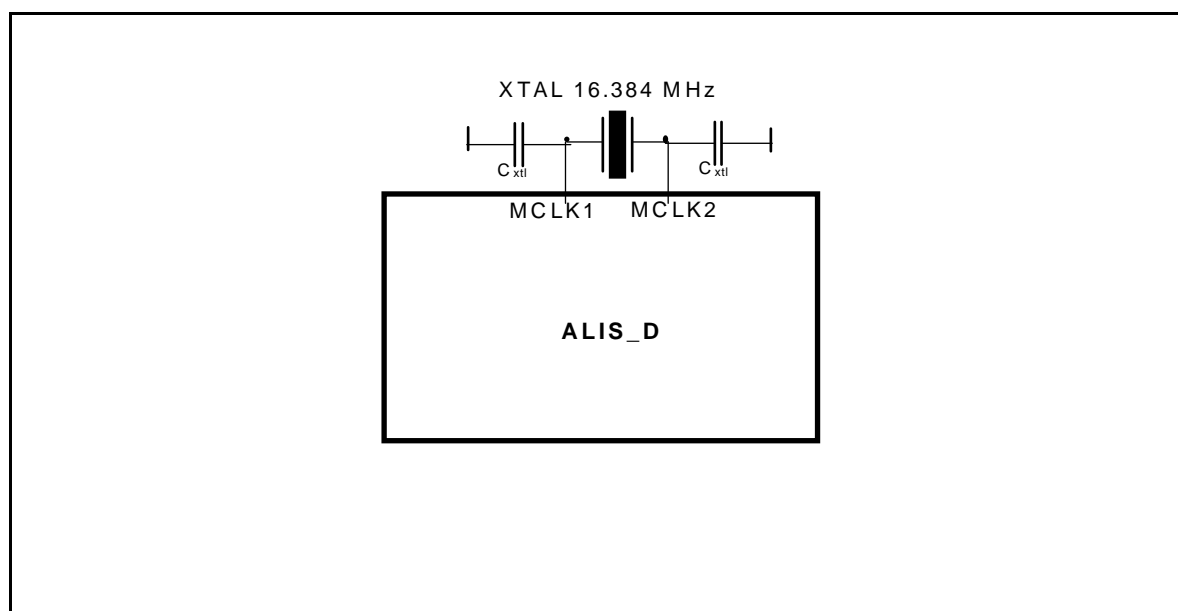


Figure 17: External crystal connections

## 5.4 Capacitive Interface

To decouple ALIS\_A from ALIS\_D, a capacitive interface is used. It is a bidirectional serial interface and used for exchanging control and data information between ALIS\_A and ALIS\_D. To avoid distortion and for performance reasons the transmission format is

## Configurations Overview

digital. For size and tolerance of the capacitors see page “ALIS Cap interface” on page 83.

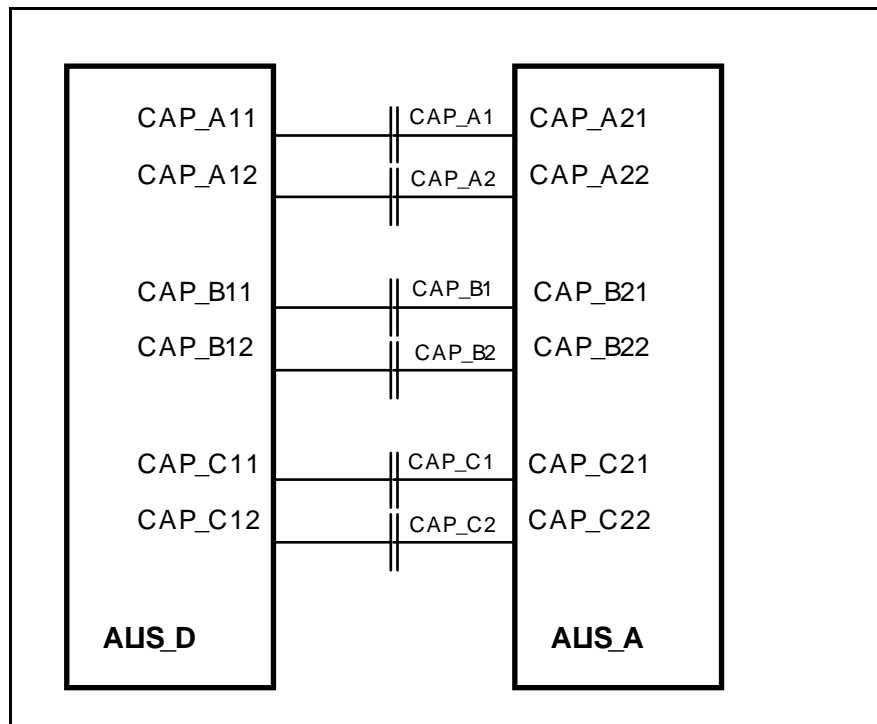


Figure 18: Connection of capacitive interface between ALIS\_A and ALIS \_D

## 5.5 Caller ID Interface

To receive the Caller ID ALIS\_D has to be connected to the line via a RC network. See "ALIS Caller ID Interface" on page 82

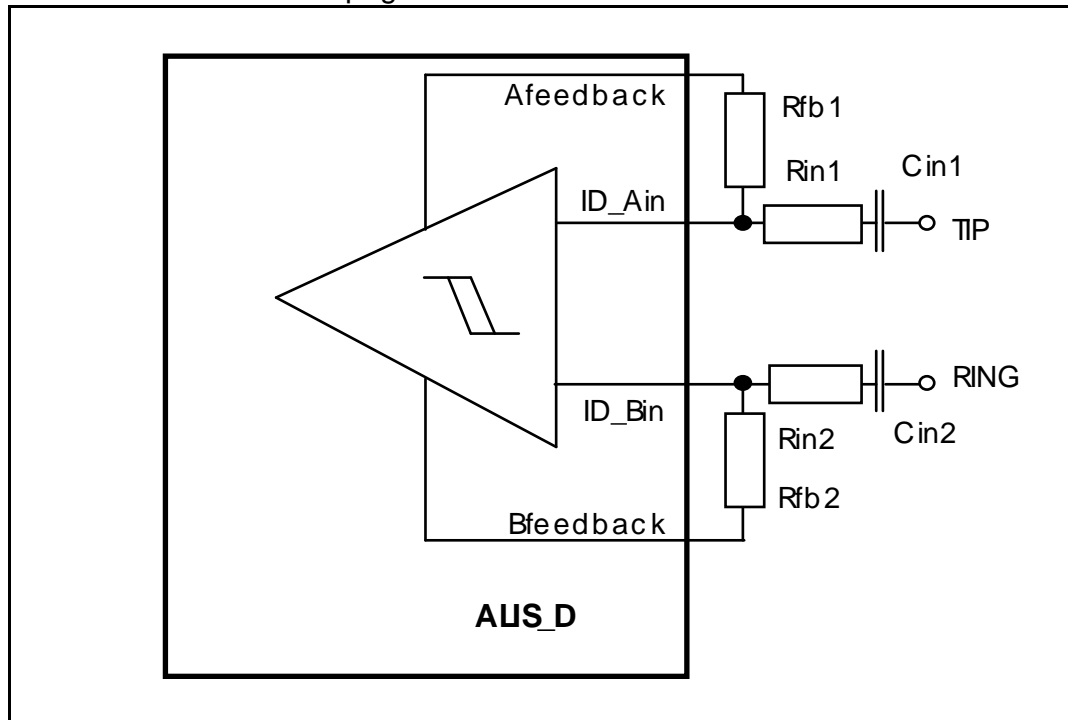


Figure 19: Caller ID interface connection of ALIS\_D to TIP/RING

## 6 Programming ALIS

With the appropriate commands ALIS can be programmed and verified very flexibly via the serial  $\mu$ -controller interface.

Four different commands are used to access the various control registers and RAMs (SOP, XOP, COP and CAO command). The first byte received via DIN, selects the command type. Each command can be used as a write or read command. Due to the extended ALIS control facilities, SOP, XOP and COP commands contain additional information for programming (writing) and verifying (reading) the ALIS status (e.g., number of subsequent bytes, software reset, operating mode).

With a SOP, XOP or COP command up to 8 bytes of data can be read or written. With the CAO command all 512 bytes of the caller-ID RAM can be read or written. Any read command causes the ALIS to respond with its specific identification byte before sending the requested information.

### 6.1 Types of Commands and Data Bytes

The ALIS commands are selected by bit 3, 4 and 6 of the command byte as shown below.

## SOP command

Bit	7	6	5	4	3	2	1	0
	PU 1	PU 0	RW	1	0	LSEL2	LSEL1	LSEL0

## XOP command

Bit	7	6	5	4	3	2	1	0
	RST	0	RW	1	1	LSEL2	LSEL1	LSEL0

## COP command

Bit	7	6	5	4	3	2	1	0
	0	0	RW	0	CODE 3	CODE 2	CODE 1	CODE 0

## CAO command

Bit	7	6	5	4	3	2	1	0
	0	1	RW	1	1	0	0	0

### 6.1.1 Storage of programming information:

- 6 Configuration registers: CR0, CR1,... CR5 accessed by SOP commands
- 8 Extended registers: XR0, XR1,... XR7 accessed by XOP commands
- 1 Coefficient RAM: CRAM accessed by COP commands
- 1 Caller ID RAM: RAM accessed by CAO commands

## 6.2 SOP Command

With the SOP-Command (status operation command) the ALIS status registers can be written or read via the  $\mu$ -controller interface.

Bit	7	6	5	4	3	2	1	0
	PU 1	PU 0	RW	1	0	LSEL2	LSEL1	LSEL0

**PU** Power Up Operation Command (only with SOP write command)

- PU = 0 0: ALIS will be set to Sleep Mode
- PU = 0 1: ALIS will be set to Ringing Mode
- PU = 1 0: ALIS will be set to Conversation Mode

PU = 1 1: ALIS will be set to Pulse Dialing Mode

**RW** Read/Write: Enables reading from the ALIS or writing information to the ALIS

RW = 0: Write to ALIS

RW = 1: Read from ALIS

**LSEL** Length select information (see also programming procedure)

This field identifies the number of subsequent data bytes

LSEL = 000: 1 byte of data is following (CR0)

LSEL = 001: 2 bytes of data are following (CR1, CR0)

LSEL = 010: 3 bytes of data are following (CR2, CR1, CR0)

LSEL = 011: 4 bytes of data are following (CR3, CR2, CR1, CR0)

LSEL = 100: 5 bytes of data are following (CR4, CR3, CR2, CR1, CR0)

LSEL = 101: 6 bytes of data are following (CR5, ..., CR1, CR0)<sup>1)</sup>

*Note: If only one configuration register requires modification, for example CR3, this can be accomplished by setting LSEL=011 and releasing pin CS after CR3 is written.*

### 6.2.1 CR0 Configuration Register 0 (filters)

Default value: 00H

Configuration register CR0 defines the basic ALIS settings, which are: enabling/disabling the programmable digital filters and tone generators.

Bit	7	6	5	4	3	2	1	0
	TH	IM	FRX	FRR	AX	AR	RIP	CLK_ EXT

**TH** Enable TH (TransHybrid Balancing)-Filter

TH = 0: TH-filter disabled

TH = 1: TH-filter enabled

**IM** Enable IM (Impedance Matching)-Filter

IM = 0: IM-filter disabled

IM = 1: IM-filter enabled

**FRX** Enable FRX (Frequency Response Transmit)-Filter

FRX = 0: FRX-filter disabled

FRX = 1: FRX-filter enabled

**FRR** Enable FRR (Frequency Response Receive)-Filter

<sup>1</sup> all other codes for testpurpose only

- FRR = 0: FRR-filter disabled  
FRR = 1: FRR-filter enabled
- AX** Enable AX (Amplification/Attenuation Transmit)-Filter  
AX = 0: AX-filter disabled  
AX = 1: AX-filter enabled
- AR** Enable AR (Amplification/Attenuation Receive)-Filter  
AR = 0: AR-filter disabled  
AR = 1: AR-filter enabled
- RIP** Enable RIP (Ringer Impedance)-Filter  
RIP = 0: RIP-filter disabled  
RIP = 1: RIP-filter enabled
- CLK\_EXT** External clock signal  
CLK\_EXT = 0: Crystal Oscillator is enabled, clock will be generated by crystal  
CLK\_EXT = 1: Crystal Oscillator is disabled, clock has to be supplied by external source

## 6.2.2 CR1 Configuration Register 1 (dialing)

Default value: 00H

Configuration register CR01 selects tone generator modes and other operation modes

Bit	7	6	5	4	3	2	1	0
	E_ Tone2	E_ Tone1	P_ Tone2	P_ Tone1	Pulse	No_ auto_ ring	RMR	RM

- E\_Tone2** Enable programmable tone generator 2  
E\_Tone2= 0: programmable tone generator 2 is disabled  
E\_Tone2= 1: programmable tone generator 2 is enabled
- E\_Tone1** Enable programmable tone generator 1  
E\_Tone1= 0: programmable tone generator 1 is disabled  
E\_Tone1= 1: programmable tone generator 1 is enabled
- P\_Tone2** User programmed frequency or fixed frequency is selected  
P\_Tone2= 0: fixed frequency for tone generator 2 is selected  
P\_Tone2= 1: programmed frequency for tone generator 2 is selected
- P\_Tone1** User programmed frequency or fixed frequency is selected  
P\_Tone1= 0: fixed frequency for tone generator 1 is selected

P\_Tone1= 1: programmed frequency for tone generator 1 is selected

**Pulse** Pulse dialing

Pulse = 1: Make for pulse dialing

Pulse = 0: Break for pulse dialing

**No\_auto\_ring**

No\_auto\_ring= 1: testmode to disable automatic switching from Sleep Mode to Ringing Mode, after valid ring.

No\_auto\_ring= 0: normal operating mode, ALIS switches automatically to ringing Mode after ring detection

**RMR** Result of Ringing Metering function (this bit can not be written)

RMR = 0: level detected was lower than the programmed<sup>1)</sup> reference

RMR = 1: level detected was higher than the programmed reference. See "Flow of Ring sequence and detection" on page 62

**RM** Ringing Metering function<sup>2)</sup>

RM = 0: Ringing metering function is disabled

RM = 1: Ringing metering function is enabled

## 6.2.3 CR2 Configuration Register 2 (caller ID)

Bit	7	6	5	4	3	2	1	0
	COT/R			IDR	Call_ pon	Call_ en	Call_I	Call_II

Default value: 00H

**COT/R** Selection of Cut Off Transmit/Receive Paths

0 0 0: Normal Operation

0 0 1: COR16 Cut Off Receive Path at 16 kHz (input of TH-Filter)

0 1 0: COR8 Cut Off Receive Path at 8 kHz

1 0 1: COT2M Cut Off Transmit Path at 2MHz (POFI-output)

1 1 0: COT64 Cut Off Transmit Path at 64 KHz (IM-filter input)

**IDR** Initialize Data RAM

IDR = 0: normal operation is selected

<sup>1</sup> The threshold can be programmed in the CRAM. Coefficients See "Ring Detect" on page 78.

<sup>2</sup> Explanation of the ringing metering function: The ring signal is rectified, and the voltage is measured. If the voltage exceeds a certain value the bit RMR is set to '1'.

IDR = 1: contents of Data RAM is set to 0 (for test purposes)

**Call\_pon** Enable the Caller ID Path

Call\_pon = 0: Caller ID Path disabled

Call\_pon = 1: Caller ID Path enabled  
(see Call\_pctl in "CR3 Configuration Register 3 (testloops)" on page 40)

**Call\_en** Enable the Caller ID

Call\_en = 1: Caller ID decoding is enabled

Call\_en = 0: Caller ID decoding is disabled

**Call\_I** Result of Caller ID decoding (this bit can not be written, test purpose only)

Call\_I = 1: 1st tone of Caller ID detected

Call\_I = 0: 1st tone of Caller ID not detected

**Call\_II** Result of Caller ID decoding (this bit can not be written, test purpose only)

Call\_II = 1: 2nd tone of Caller ID detected

Call\_II = 0: 2nd tone of Caller ID not detected

## 6.2.4 CR3 Configuration Register 3 (testloops)

Bit	7	6	5	4	3	2	1	0
	Test-Loops				SEL	Call_pctl	DHP-R	DHP-X

Default value: 00H

**Test-Loops** 4 bit field for selection of analog and digital loop backs

0101	ALB_CIF: Cap. interface loop the signal from the input circuit (CAP_B11/12 is connected to the output drivers (CAP_A11/12, CAP_C11/12)
1000	ALB-CID: Caller ID loop; the output signal from the caller ID comparator is connected to the output drivers of the capacative interface (CAP_A11/12, CAP_C11/12);
1001	DLB-2M: Loop via HW-filters;
1100	DLB-128k: Loop inside DSP;
1101	DLB-64k: Loop inside DSP;
1111	DLB-PCM: Loop via PCM-interface; the received data is sent back in the next frame;

**SEL** Test loop selection



SEL = 0: Test loops via impedance path are selected

SEL = 1: Test loops via receive path are selected

**Call\_pctl** Caller ID Path control

Call\_pctl = 0: Caller ID interface is enabled during ringing mode

Call\_pctl = 1: Caller ID interface will be selected by the Call\_pon bit in CR2

*Note: For test purposes the path can be controlled manually. Must be '0' for normal operation.*

**DHP-X** Disable highpass in transmit direction

DHP-X = 0: transmit high pass is enabled

DHP-X = 1: transmit high pass is disabled

**DHP-R** Disable highpass in receive direction

DHP-R = 0: receive high pass is enabled

DHP-R = 1: receive high pass is disabled

## 6.2.5 CR4 Configuration Register 4 (analog gain)

Bit	7	6	5	4	3	2	1	0
	AGR_Z 1	AGR_Z 0	AGR_R 1	AGR_R 0	AGX 1	AGX 0	Int_en	Fsc_en

Default value: 00H

**AGR\_Z** Analog gain in impedance loop (can be used as AGC)

AGR\_Z = 00: analog gain A is disabled (0 dB amplification)

AGR\_Z = 11: analog gain A is enabled (2.5 dB amplification)

AGR\_Z = 10: analog gain A is enabled (6 dB amplification)

AGR\_Z = 01: analog gain A is enabled (-3.5 dB amplification)<sup>1)</sup>

**AGR\_R** Analog gain in receive direction (can be used as AGC)

AGR\_R = 00: analog gain B is disabled (0 dB amplification)

AGR\_R = 01: analog gain B is enabled (3.5 dB amplification)

AGR\_R = 11: analog gain B is enabled (6 dB amplification)

**AGX** Analog gain in transmit direction (can be used as AGC)

AGX = 00: analog gain A is disabled (0 dB amplification)

<sup>1</sup> Note: for stability reasons the sum of AGR\_Z and AGX should be zero.

AGX = 01: analog gain A is enabled (-6 dB amplification)  
 AGX = 10: analog gain A is enabled (3.5 dB amplification)  
 AGX = 11: analog gain A is enabled (-2.5 dB amplification)

**Int\_en** Interrupt enable

Int\_en = 1: Enable interrupts

Int\_en = 0: Disable interrupts

**Fsc\_en** FSC signal source selection

Fsc\_en = 0: FSC must be generated externally

Fsc\_en = 1 FSC is generated internally

## 6.2.6 CR5 Configuration Register 5 (Version)

Bit	7	6	5	4	3	2	1	0
	V_7	V_6	V_5	V_4	V_3	V_2	V_1	V_0

**V** The actual version of the ALIS (this byte can not be written)  
 02H for ALIS V2.1

## 6.3 XOP Command

With the XOP command (extended operation command) the ALIS digital command/indication interface to the line and external equipment is configured and evaluated. Also other common functions are assigned with this command.

Bit	7	6	5	4	3	2	1	0
	RST	0	RW	1	1	LSEL2	LSEL1	LSEL0

**RST** Software Reset (same as RESET-pin)

RST = 0: No reset

RST = 1: ALIS is reset to the default settings

**RW** Read / Write: Enables reading from the ALIS or writing to the ALIS

RW = 0: Write to ALIS

RW = 1: Read from ALIS

**LSEL** Length select information, specifies the number of subsequent data bytes

LSEL = 000 1 byte of data is following (XR0)

LSEL = 001      2 bytes of data are following (XR1, XR0)

:

LSEL= 111      8 bytes of data are following (XR7, ..., XR1, XR0)

### 6.3.1 XR0 Extended Register 0 (Interrupt Register)

Any interrupt indications can be monitored in the interrupt register. Interrupts can be signalled through a logic high at the INT-line. After an indication occurred further loading of the interrupt register is locked until its contents is read via the  $\mu$ -Controller interface. Reading of the interrupt register XR0 releases the lock and the INT-line is set to low again. See "Interrupt Controller" on page 58 for more details.

#### In connection with XOP-Read commands

Bit	7	6	5	4	3	2	1	0
	0	Wake_up	Cadence	RING	Caller_ID	VDD_OK	SI_1	SI_0

Default value: 00H

#### Wake\_up      Wake\_up Interrupt

Wake\_up = 0: no Wake\_up Interrupt

Wake\_up = 1: if CLK\_OFF bit is set (See "XR6 Extended Register 6 (Power State)" on page 47) and a ring signal occurs<sup>1)</sup> a Wake\_up Interrupt is generated. To clear this interrupt the CLK\_OFF bit must be reset and ALIS\_D has to be supplied with a clock.

#### Cadence      Cadence Interrupt

Cadence = 0: no Cadence Interrupt  
(time between two ring bursts is available in XR4)

Cadence = 1: time between two ring bursts exceeds the programmed time  
(see "XR2 Extended Register 2 (Cadence Time Out)" on page 45)

#### RING      Ring Interrupt

RING = 0: no Ring Burst

RING = 1: No\_auto\_ring= 0: this bit is set after the second valid Ring Burst  
No\_auto\_ring= 1: ALIS stays in the Sleep Mode and waits for a command. This bit represents the Ring detection signal from the ALIS\_A. See "CR1 Configuration Register 1 (dialing)" on page 38

<sup>1</sup> any signal at the line with a voltage of more than 18 V. To decode a valid ring signal ALIS must be switched to the Ringing Mode.

*Note: In this case a command is mandatory to avoid a deadlock.*

**Caller\_ID** Caller ID Interrupt

Caller\_ID = 0: no Caller ID preamble detected

Caller\_ID = 1: Caller ID preamble detected

**VDD\_OK** Vdd at ALIS\_A Interrupt

VDD\_OK = 1: power supply for ALIS\_A is available and connection between ALIS\_A and ALIS\_D is working

VDD\_OK = 0: no power supply for ALIS\_A or no connection between ALIS\_A and ALIS\_D

**SI\_0** status of pin SI\_0 at ALIS\_A is transferred to this register

**SI\_1** status of pin SI\_1 at ALIS\_A is transferred to this register

*Note: The auxiliary pins (SO\_0, SO\_1, SI\_0, SI\_1) are isolated via the capacitive interface.*

**In connection with XOP-Write commands to control the Output pins SO**

Bit	7	6	5	4	3	2	1	0
	0	0	0	0	0	SO_2	SO_1	SO_0

**SO\_0** pin SO\_0 at ALIS\_A is set to the assigned value if ALIS is not in sleep mode

**SO\_1** pin SO\_1Q at ALIS\_A is set to the inverted assigned value if ALIS is not in sleep mode

**SO\_2** pin SO at ALIS\_D is set to the assigned value

### 6.3.2 XR1 Extended Register 1 (Interrupt enable Register)

Bit	7	6	5	4	3	2	1	0
	0	M_Wake_up	M_Cadence	M_RING	M_Caller_ID	M_VDD_OK	M_SI_1	M_SI_0

Default value: 00H

**M\_Wake\_up**

M\_Wake\_up = 0:Disable Wake\_up Interrupt

M\_Wake\_up = 1:Enable Wake\_up Interrupt

## M\_Cadence

M\_Cadence = 0:Disable Cadence Interrupt

M\_Cadence = 1:Enable Cadence Interrupt

## M\_RING

M\_RING = 0: Disable RING Interrupts

M\_RING = 1: Enable RING Interrupts

## M\_Caller\_ID

M\_Caller\_ID = 0:Disable Caller\_ID Interrupt

M\_Caller\_ID = 1:Enable Caller\_ID Interrupt

## M\_VDD\_OK

M\_VDD\_OK = 0:Disable VDD Interrupt

M\_VDD\_OK = 1:Enable VDD Interrupt

## M\_SI\_1

M\_SI\_1 = 0: Disable Interrupts of SI\_1

M\_SI\_1 = 1: Enable Interrupts of SI\_1

## M\_SI\_0

M\_SI\_0 = 0: Disable Interrupts of SI\_0

M\_SI\_0 = 1: Enable Interrupts of SI\_0

### 6.3.3 XR2 Extended Register 2 (Cadence Time Out)

Bit	7	6	5	4	3	2	1	0
	CTO 7	CTO 6	CTO 5	CTO 4	CTO 3	CTO 2	CTO 1	CTO 0

Default value: 7DH

## CTO ms

Programmable Cadence Time Out: if the time between the first two ring bursts exceeds the time programmed in this register a cadence interrupt is generated. The time out is programmable in steps of 64 ms up to 16 seconds.

*Note: 00 means no Cadence timeout programmed - no interrupt will be generated.*

## 6.3.4 XR3 Extended Register 3 (DC Characteristic)

Bit	7	6	5	4	3	2	1	0
	AGB1	AGB0	B_off	DCU 1	DCU 0	DCI	DCR 1	DCR 0

**AGB** analog gain for the analog transhybrid filter

AGB = 00: Gain for analog transhybrid - filter = 24.6 dB

AGB = 01: Gain for analog transhybrid - filter = 21.6 dB

AGB = 10: Gain for analog transhybrid - filter = 18.6 dB

AGB = 11: Gain for analog transhybrid - filter = 16.1 dB

**B\_off** Enable analog Transhybrid filter

B\_off = 0: analog transhybrid - filter on

B\_off = 1: analog transhybrid - filter off

*Note: the analog transhybrid filter is a analog Prefilter optimised for long loops with a transhybrid loss of about 10 dB.*

DCU = 00: U0 for DC Characteristic is 0 V

DCU = 01: U0 for DC Characteristic is 1.5 V

DCU = 10: U0 for DC Characteristic is 3.5 V

DCU = 11: U0 for DC Characteristic is 7.2 V

*Note: these values does not include the voltage drop at the external diodes see also "DC Characteristics" on page 80*

**DCI** Limit current for the DC characteristic

DCI = 0: Limit current is 100mA

DCI = 1: Limit current is 50 mA

**DCR** Resistance of the DC characteristic

DCR = 00: R for DC Characteristic is 280  $\Omega$

DCR = 01: R for DC Characteristic is 240  $\Omega$

DCR = 10: R for DC Characteristic is 200  $\Omega$

DCR = 11: R for DC Characteristic is 100  $\Omega$

*Note: If DCU is programmed to 7.2V (DCU = 11) R for DC Characteristic is always 70  $\Omega$  independent of the contents of DCR. See "DC Termination" on page 80*

## 6.3.5 XR4 Extended Register 4 (Cadence)

Bit	7	6	5	4	3	2	1	0
	C_7	C_6	C_5	C_4	C_3	C_2	C_1	C_0

**C** ms (read only)

Contains the measured time between the two first ring bursts (time step 64 ms) if the time is below the cadence time-out as programmed in XR2.

## 6.3.6 XR5 Extended Register 5 (Ring timer)

Bit	7	6	5	4	3	2	1	0
	T_7	T_6	T_5	T_4	T_3	T_2	T_1	T_0

Default value: 22H

**T** ms

Ring latency timer, programmable in steps of 2 ms

ALIS\_A decodes any signal with more than 18 V at TIP/RING. This signal will be handled over to ALIS\_D for further processing. This timer bridges the time when the sinewave of the ringsignal is below the noted 18 V to make sure that ALIS does not fall back into the sleep mode.

## 6.3.7 XR6 Extended Register 6 (Power State)

Bit	7	6	5	4	3	2	1	0
	0	0	0	CKL_ OFF	0	0	CPS1	CPS0

Default value: 00H

**CPS** Current Power State (read only)

CPS = 00 Power state is sleep

CPS = 01	Power state is ringing
CPS = 10	Power state is conversation
CPS = 11	Power state is pulse dialing

*Note: the power mode can be programmed by the SOP command. In this register the current power state will be reflected.*

**CLK\_OFF** Turn off master clock (ALIS is programmed to deep- sleep mode)

CLK\_OFF = 0 Master clock is not turned off internally

CLK\_OFF = 1 Master clock is turned off internally

*Note: the external clock can be turned off after setting the CLK\_OFF bit. For programming ALIS the clock must be switched on.*

*Note: when a crystal is used it will be turned off automatically when the CLK\_OFF bit is set. It will be switched on when the CS signal goes low. However the user must wait until the crystal is working before initiating a command.*

### 6.3.8 XR7 Extended Register 7 (Vdd)

Bit	7	6	5	4	3	2	1	0
	0	0	0	0	0	Vdd1	Vdd0	0

Default value: 00H

**Vdd** Current Power State

Vdd = 00 Vdd of ALIS\_A is 4.25 V

Vdd = 01 Vdd of ALIS\_A is 4.4 V (Test purpose only)

Vdd = 10 Vdd of ALIS\_A is 3.85 V (Test purpose only)

Vdd = 11 Vdd of ALIS\_A is 4 V

### 6.4 COP Command

With a COP Command the coefficients for the programmable filters can be written to ALIS Coefficient RAM or read from the Coefficient RAM via the  $\mu$ -Controller interface for verification.



Bit	7	6	5	4	3	2	1	0
	0	0	RW	0	CODE 3	CODE 2	CODE 1	CODE 0

**RW** Read/Write

RW = 0 Subsequent data is written to ALIS

RW = 1 Read data from ALIS

**CODE** includes number of following bytes and filter-address

0	0	0	0	TH-Filter coefficients (part 1)	(followed by 8 bytes of data)
0	0	0	1	TH-Filter coefficients (part 2)	(followed by 8 bytes of data)
0	0	1	0	TH-Filter coefficients (part 3)	(followed by 8 bytes of data)
0	0	1	1	Ringer Impedance (part 1)	(followed by 8 bytes of data)
0	1	0	0	IM-Filter coefficients (part 1)	(followed by 8 bytes of data)
0	1	0	1	IM-Filter coefficients (part 2)	(followed by 8 bytes of data)
0	1	1	0	Ringer Impedance (part 2)	(followed by 8 bytes of data)
0	1	1	1	FRR-Filter coefficients	(followed by 8 bytes of data)
1	0	0	0	FRX-Filter coefficients	(followed by 8 bytes of data)
1	0	0	1	AR-Filter coefficients	(followed by 4 bytes of data)
1	0	1	0	AX-Filter coefficients	(followed by 4 bytes of data)
1	0	1	1	Tone1- coefficients	(followed by 4 bytes of data)
1	1	0	0	Tone2- coefficients	(followed by 4 bytes of data)
1	1	0	1	Levelmetering Ringing	(followed by 4 bytes of data)
1	1	1	0	Caller ID 1st Tone	(followed by 8 bytes of data)
1	1	1	1	Caller ID 2nd Tone	(followed by 8 bytes of data)

## 6.5 CAO Command

With a CAO Command the decoded Caller ID can be read. A CAO Command is always followed by 512 bytes of data.

Bit	7	6	5	4	3	2	1	0
	0	1	RW	1	1	0	0	0

**RW** Read/Write

RW = 0 Subsequent data is written to ALIS (test purpose only)

RW = 1 Read data from ALIS

## 6.6 Register Summary

### 6.6.1 CR Registers:

Bit	7	6	5	4	3	2	1	0
CR0	TH	IM	FRX	FRR	AX	AR	RIP	CLK_EXT
CR1	E_Tone2	E_Tone1	P_Tone2	P_Tone1	Pulse	No_auto	RMR	RM
CR2	COT/R			IDR	Call_p on	Call_e n	Call_I	Call_II
CR3	TestLoops				SEL	Cal _pctl	DHP-R	DHP-X
CR4	AGR_ Z 1	AGR_ Z 0	AGR_ R1	AGR_ R 0	AGX 1	AGX 0	Int_en	Fsc_ en
CR5	V_7	V_6	V_5	V_4	V_3	V_2	V_1	V_0

Table 5: Summary of CR Registers

## 6.6.2 XR Registers:

Bit	7	6	5	4	3	2	1	0
XR0/R	0	Wake_up	Cadence	RING	Caller_ID	VDD_OK	SI_1	SI_0
XR0/W	0	0	0	0	0	SO_2	SO_1	SO_0
XR1	0	M_Wake_up	M_Cadence	M_RING	M_Caller_ID	M_VDD_OK	M_SI_1	M_SI_0
XR2	CTO 7	CTO 6	CTO 5	CTO 4	CTO 3	CTO 2	CTO 1	CTO 0
XR3	AGB1	AGB0	B_off	DCU 1	DCU 0	DCI	DCR 1	DCR0
XR4	C_7	C_6	C_5	C_4	C_3	C_2	C_1	C_0
XR5	T_7	T_6	T_5	T_4	T_3	T_2	T_1	T_0
XR6	0	0	0	CLK_OFF	0	0	CPS1	CPS0
XR7	0	0	0	0	0	Vdd1	Vdd0	0

Table 6: Summary of CR Registers

## 7 ALIS Command Structure

### 7.1 SOP Commands

#### 7.1.1 SOP - Write Commands

DIN	7	6	5	4	3	2	1	0	Bit	7	6	5	4	3	2	1	0	DOUT
SOP-Write 1 Byte	x	x	0	1	0	0	0	0		Idle								
CR0	Data									Idle								

DIN	7	6	5	4	3	2	1	0	Bit	7	6	5	4	3	2	1	0	DOUT
SOP-Write 2 Bytes	x	x	0	1	0	0	0	1		Idle								
CR1	Data									Idle								
CR0	Data									Idle								

## ALIS Command Structure

DIN	7	6	5	4	3	2	1	0	Bit	7	6	5	4	3	2	1	0	DOUT
SOP-Write 3 Bytes	x	x	0	1	0	0	1	0										Idle
CR2																		Idle
CR1																		Idle
CR0																		Idle

DIN	7	6	5	4	3	2	1	0	Bit	7	6	5	4	3	2	1	0	DOUT
SOP-Write 4 Bytes	x	x	0	1	0	0	1	1										Idle
CR3																		Idle
CR2																		Idle
CR1																		Idle
CR0																		Idle

### 7.1.2 SOP - Read Commands

DIN	7	6	5	4	3	2	1	0	Bit	7	6	5	4	3	2	1	0	DOUT
SOP-Read 1 Byte	x	x	1	1	0	0	0	0										Idle
																		1 0 0 0 0 0 0 1 ID
																		Data CR0

DIN	7	6	5	4	3	2	1	0	Bit	7	6	5	4	3	2	1	0	DOUT
SOP-Read 2 Bytes	x	x	1	1	0	0	0	1										Idle
																		1 0 0 0 0 0 0 1 ID
																		Data CR1
																		Data CR0

DIN	7	6	5	4	3	2	1	0	Bit	7	6	5	4	3	2	1	0	DOUT
SOP-Read 3 Bytes	x	x	1	1	0	0	1	0										Idle
																		1 0 0 0 0 0 0 1 ID
																		Data CR2

## ALIS Command Structure

Idle
Idle

Data	CR1
Data	CR0

DIN								Bit									DOUT							
SOP-Read 4 Bytes								x	x	1	1	0	0	1	1									
								Idle								Idle								
								Idle								ID								
								Data								CR3								
								Data								CR2								
								Data								CR1								
								Data								CR0								

*Note: x according to the description of the Power Up operation command see “SOP Command” on page 36*

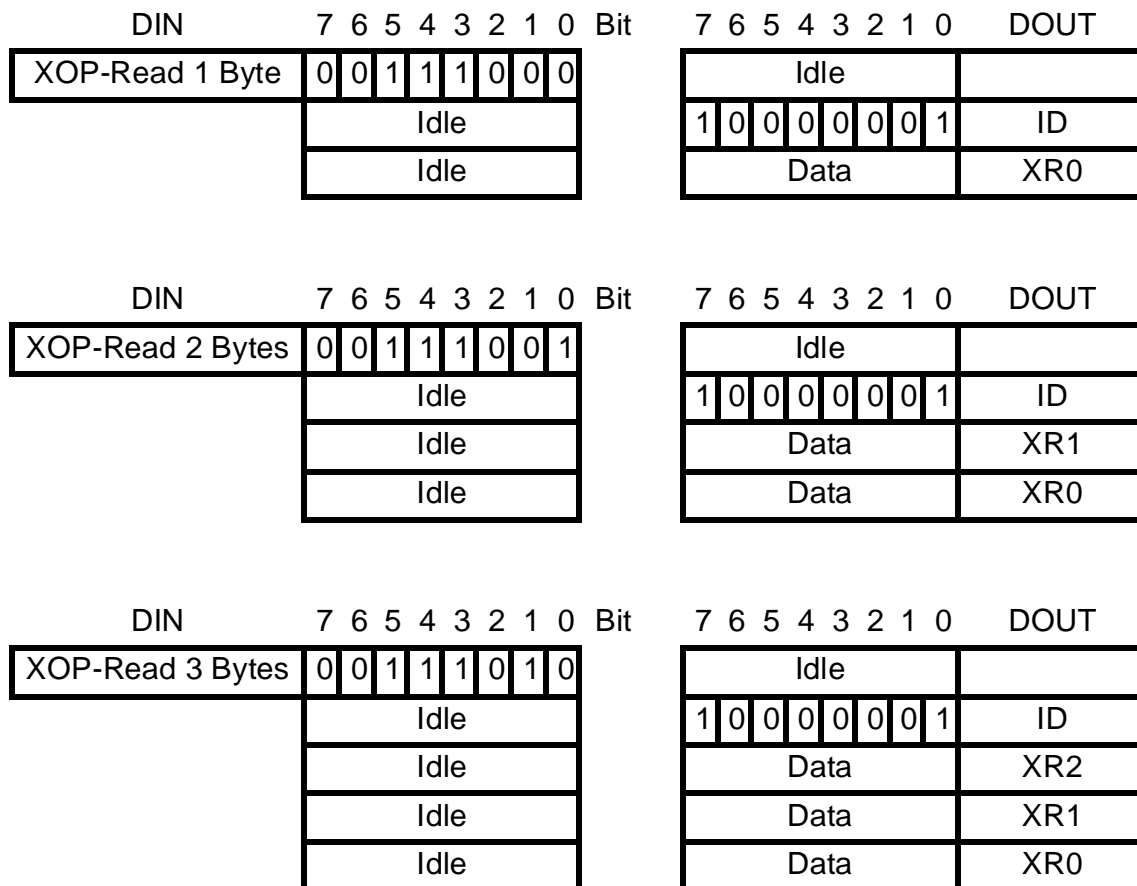
## 7.2 XOP Commands

### 7.2.1 XOP - Write Commands

DIN	7	6	5	4	3	2	1	0	Bit	7	6	5	4	3	2	1	0	DOUT
XOP-Write 2 Bytes	0	0	0	1	1	0	0	1		Idle								
XR1	Data									Idle								
XR0	Data									Idle								

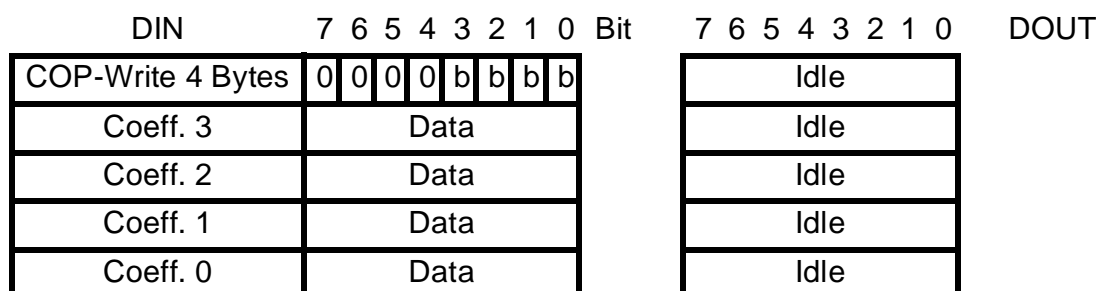
DIN	7	6	5	4	3	2	1	0	Bit	7	6	5	4	3	2	1	0	DOUT
XOP-Write 3 Bytes	0	0	0	1	1	0	1	0		Idle								
XR2	Data									Idle								
XR1	Data									Idle								
XR0	Data									Idle								

## 7.2.2 XOP - Read Commands



## 7.3 COP Command

### 7.3.1 COP - Write Commands



DIN		7	6	5	4	3	2	1	0	Bit	7	6	5	4	3	2	1	0	DOUT
COP-Write 8 Bytes		0	0	0	0	b	b	b	b		Idle								
Coeff. 7		Data									Idle								
Coeff. 6		Data									Idle								
Coeff. 5		Data									Idle								
Coeff. 4		Data									Idle								
Coeff. 3		Data									Idle								
Coeff. 2		Data									Idle								
Coeff. 1		Data									Idle								
Coeff. 0		Data									Idle								

### 7.3.2 COP - Read Commands

DIN	7	6	5	4	3	2	1	0	Bit	7	6	5	4	3	2	1	0	DOUT
COP-Read 4 bytes	0	0	1	0	b	b	b	b		Idle								
										Idle								
										Idle								
										Idle								
										Idle								
										Idle								
										1	0	0	0	0	0	0	1	ID
										Data								Coeff.3
										Data								Coeff.2
										Data								Coeff.1
										Data								Coeff.0

DIN	7	6	5	4	3	2	1	0	Bit	7	6	5	4	3	2	1	0	DOUT
COP-Read 8 bytes	0	0	1	0	b	b	b	b		Idle								
	Idle									1	0	0	0	0	0	0	1	ID
	Idle									Data								Coeff.7
	Idle									Data								Coeff.6
	Idle									Data								Coeff.5
	Idle									Data								Coeff.4
	Idle									Data								Coeff.3
	Idle									Data								Coeff.2

Idle
Idle

Data	Coeff.1
Data	Coeff.0

## 7.4 CAO Command

DIN	7	6	5	4	3	2	1	0	Bit
CAO Write	0	1	0	1	1	0	0	0	
	Caller ID 512								
	Caller ID 511								
	Caller ID 510								
	...								
	Caller ID 1								

DIN	7	6	5	4	3	2	1	0	Bit
CAO Read	0	1	1	1	1	0	0	0	
	Idle								
	Idle								
	Idle								
	Idle								
	Idle								

7	6	5	4	3	2	1	0	DOUT
1	0	0	0	0	0	0	1	ID
Data								Caller ID 512
Data								Caller ID 511
								...
Data								Caller ID 1



### 7.5 Example for a Mixed Command

Every single command must begin with a falling edge of CS.

DIN	7	6	5	4	3	2	1	0	Bit	7	6	5	4	3	2	1	0	DOUT
SOP-Write 4 Bytes	x	x	0	1	0	0	1	1										Idle
CR3																		Idle
CR2																		Idle
CR1																		Idle
CR0																		Idle
XOP-Write 2 Bytes	0	0	0	1	1	0	0	1										Idle
XR1																		Idle
XR0																		Idle
COP-Write 4 Bytes	0	0	0	0	b	b	b	b										Idle
Coeff. 3																		Idle
Coeff. 2																		Idle
Coeff. 1																		Idle
Coeff. 0																		Idle
SOP-Read 3 Bytes	x	x	1	1	0	0	1	0										Idle
										1	0	0	0	0	0	0	1	ID
																		Data
																		CR2
																		Data
																		CR1
																		Data
																		CR0
																		Idle
COP-Read 4 Bytes	0	0	1	0	b	b	b	b		1	0	0	0	0	0	0	1	ID
																		Data
																		Coeff.3
																		Data
																		Coeff.2
																		Data
																		Coeff.1
																		Data
																		Coeff.0
																		Idle
XOP-Read 1 Byte	0	0	1	1	1	0	0	0		1	0	0	0	0	0	0	1	ID
																		Data
																		XR0

## **8 Interrupt Controller**

There are seven different sources that can cause interrupts in ALIS. The status of these sources can be read in the interrupt- register XR0. Every interrupt source can be enabled individually in the interrupt enable register XR1.

For monitoring an interrupt source the corresponding bit must be set in XR1. The interrupt register is locked if an enabled interrupt- indication occurs. The lock is released by reading the interrupt- register. Any interrupt indication that occurs during a lock- period will be detected after releasing the lock.

The INT-pad can be used as indication for external hardware to read the interrupt register. If the Int\_en bit (CR4) is set, the INT-pad goes to high whenever the interrupt register is locked.

The host must analyse the bits in the interrupt- register to determine the cause for the pending interrupt. All interrupt- sources that are not enabled must be ignored by the host for analysis. It is possible that several sources cause only one interrupt all together! (i.e. break- down of serial connection to ALIS\_A: VDD\_OK, SI\_0, SI\_1; if more interrupts occur during the lock- period). If the interrupt was caused by a CADENCE, RING, CALLER\_ID or WAKE\_UP interrupt the indication that caused the pending interrupt is reset by reading the interrupt register XR0.

As only one interrupt can be stored internally the host must respond immediately to avoid loss of interrupts.

### **8.1 Nature and sources of interrupts:**

There are three different kinds of interrupt indications depending on their source:

## 8.1.1 Interrupt indication on signal change:

Interrupts:	SI_0, SI_1, VDD_OK;
Sources:	Signalling pins at ALIS_A (SI_0, SI_1); VDD_OK indicates that ALIS_A has power supply and that the serial connection via the cap. interface is working;
Interrupt indication:	Any change of the signals will generate an interrupt. The host must store the previous state of these bits to check which signal caused an interrupt.
Note:	These bits will go to '0' when there is no connection to ALIS_A via the cap. interface, which will cause interrupts!
Lock behaviour:	At <i>lock-release-time</i> the current signal is compared to the signal stored at <i>lock-time</i> . Any difference will cause another interrupt.

## 8.1.2 Interrupt indication on event:

Interrupts:	CALLER_ID, RING, CADENCE;
Sources:	CALLER_ID: complete marker sequence of caller ID detected; RING: depending on automatic mode-switching: - detection of more than 18 V at TIP/RING (No_auto_ring '1'); - 2nd valid ringing (No_auto_ring '0'); CADENCE: time-out for 2nd ring- burst; the time can be programmed in XR2 (No_auto_ring '0');
Interrupt indication:	These interrupts indicate that a certain event has happened. The bits are set from their source and can only be reset from the host by reading the interrupt register. Whenever one of these bits is set, this is an indication that this event has happened.
Lock behaviour:	If one of these events occurs during the register is locked another interrupt will be generated as soon as the lock is released.

### 8.1.3 Interrupt indication on high level:

If ALIS\_D is set to the deep-sleep mode (XR6, CLK\_OFF = '1') this interrupt indicates that there is a signal greater than 18 V at TIP/RING.

Interrupts:	WAKE_UP;
Source:	ring_detect signal from ALIS_A;
Interrupt indication:	more than 18 V at TIP/RING. It is cleared after reading the interrupt- register. Another interrupt is generated if the signal remains higher than 18 V.
Lock behaviour:	The interrupt will lock the register as soon as clock is turned on again! (If no clock signal is applied to ALIS_D the other interrupts cannot occur anyway.)

## 9 Operating Modes

### 9.1 Reset (Basic setting mode)

Condition: RESET low, MCLK can be down

ALIS \_D:

After initial application of VDD or setting pin RESET to '0' during operation, or by software-reset (see XOP command), ALIS \_D enters a basic setting mode. Basic setting means, that ALIS\_ D configuration registers CR0... CR5 and XR0... XR7 are initialized to the default value (Sleep mode). All programmable filters are disabled.

If any voltage is applied to any input-pin before initial application of VDD, ALIS may not enter the basic setting mode. In this case it is necessary to reset ALIS or to initialize ALIS configuration registers to the default value.

ALIS \_A:

When the plug is connected to TIP / RING and the hook switch is closed, ALIS \_A generates its supply voltage out of the line current and does a power on reset.

### 9.2 Deep Sleep Mode

Condition: RESET high, MCLK can be switched off, CLK\_OFF bit is set

Will be entered from any mode by programming the CLK\_OFF bit in XR6. During The Deep Sleep mode the serial control interface is ready to receive and register commands only when the MCLK is switched on. See "XR6 Extended Register 6 (Power State)" on page 47. Incoming rings will be indicated by the Wake\_up Interrupt.

### 9.3 Sleep Mode

Condition: RESET high, MCLK must run

After releasing the RESET-pin (RESET-state), ALIS will enter the Sleep Mode. ALIS is forced to Sleep Mode with the PU (power up) bits set to '00' in the SOP command. During Sleep Mode the serial control interface is ready to receive commands and transmit data. Received voice data on DAT\_IN-pin will be ignored. ALIS configuration registers Caller ID Ram and coefficient Ram can be loaded and read back in this mode.

#### **9.4 Ringing Mode**

Condition: RESET high, MCLK must run

Will be entered automatically when bit No\_auto\_ring is set to 0 from Sleep mode after first ring pulse or with the PU bits set to '01' in the SOP command. In this mode ALIS will measure level, frequency and cadence of the ring signal. The cadence between the first two ring bursts is stored in XR4. If the bit Caller\_en is enabled an incoming caller ID will be decoded and stored (see CAO command).

#### **9.5 Conversation Mode**

Condition: RESET high, MCLK must run

The operating mode is entered upon recognition of the PU bits set to '10' in a SOP command.

In Conversation mode AC impedance and the DC - Loops are switched on. The programmed AC and DC characteristics are realized by this loops. Receive and transmit path are on. The tonegenerators are available.

#### **9.6 Pulse Dialing Mode**

Condition: RESET high, MCLK must run, off-hook

The Pulse Dialing Mode is entered by setting the PU bits to '11' in a SOP command.

In Pulse Dialing Mode the external transistor T1 will be switched on and off according to the bit PULSE in CR1. The pulse timing must be controlled by the host.

## 9.7 Operating Flow Chart

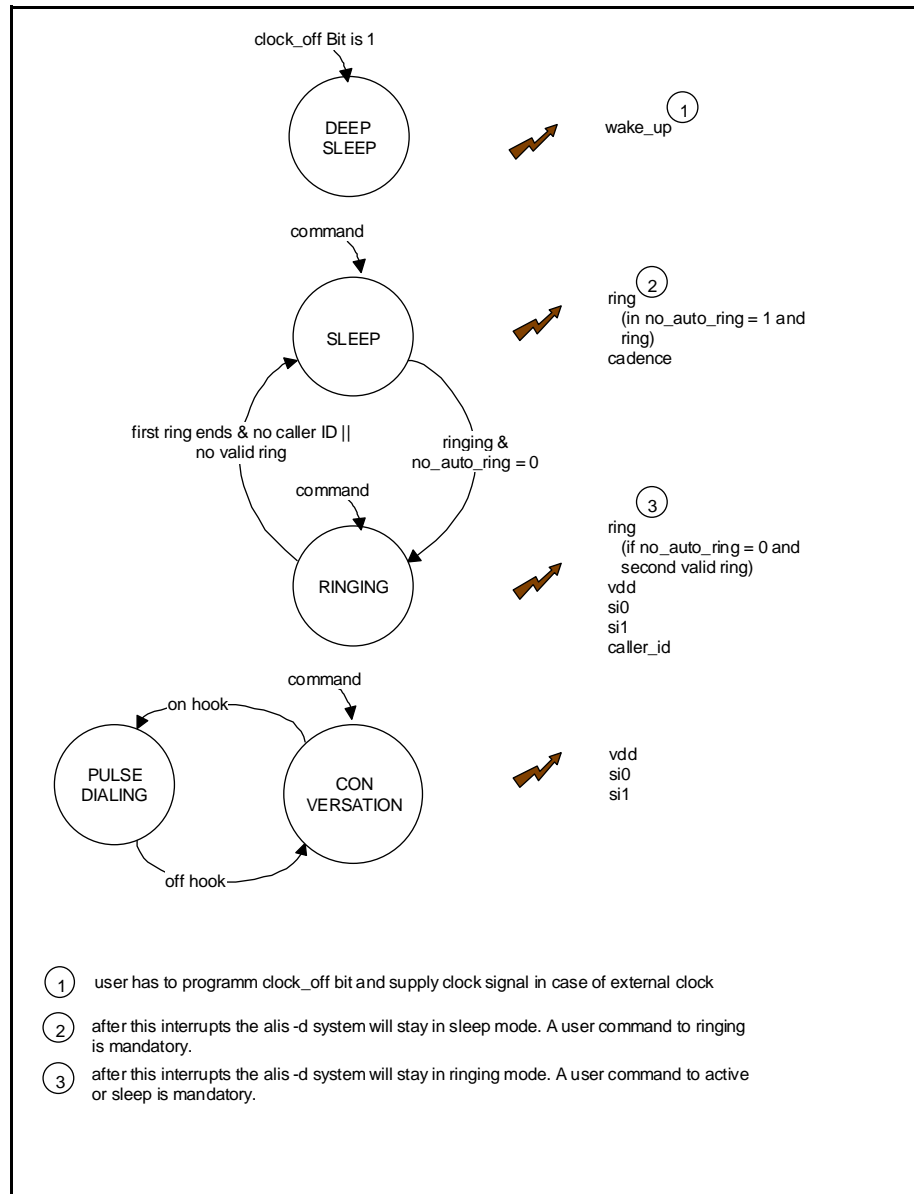


Figure 20: Operation modes transitions and Interrupts

## 9.8 Flow of Ring sequence and detection

Ring detection works in ALIS as a two step procedure.

As a first step ALIS\_A will detect any AC signal at TIP and RING with a peak value of more than 18 V and generate the signal Ring\_detect. This signal can either generate an interrupt or switch ALIS into the Ringing Mode depending on the No\_auto\_ring bit in CR1 (See “CR1 Configuration Register 1 (dialing)” on page 38) The current power mode can be read in “XR6 Extended Register 6 (Power State)” on page 47.

As a second step only if enabled by RM (See “CR1 Configuration Register 1 (dialing)” on page 38) in the Ringing Mode the TIP/RING Signal will be bandfiltered and compared to a programmable threshold. If the result is higher than this threshold the signal RMR-bit is set to one. Ring-threshold can be polled as RMR-bit in CR1. The following flow charts show these sequences in more detail.

*Note: The initial connection to TIP RING looks like a ring voltage to ALIS\_A. ALIS\_D reaction depends on auto\_ring bit:*

- a) auto\_ring: ALIS\_D goes to ringing, since spike is no valid ringsignal after that ALIS\_D goes to sleep mode.*
- b) No\_auto\_ring: ALIS\_D generates a ring interrupt, stays in sleep mode and waits for a command. The user has to switch ALIS\_D to ringing and poll the RMR bit. If ringing is not valid (RMR bit = 0) the chip can be set back to sleep mode.*

To detect a valid ring signal and Caller ID ALIS automatically must be programmed to the following setting:

- set RM to '1' “CR1 Configuration Register 1 (dialing)” on page 38
- cadence timeout must be programmed to PTT requirements “XR2 Extended Register 2 (Cadence Time Out)” on page 45
- ring latency timer must be programmed to a value higher four times of the ring period “XR5 Extended Register 5 (Ring timer)” on page 47
- valid ring coefficients
- set No\_auto\_ring to '0' “CR1 Configuration Register 1 (dialing)” on page 38
- enable Call\_en “CR2 Configuration Register 2 (caller ID)” on page 39
- enable corresponding interrupts

## 9.8.1 Successful Ring\_sequence, auto Ring enabled, No Caller ID

The following chart and diagram shows a successful flow of a ring-event detection with automatic power mode change (No\_auto\_ring = 0, Caller\_en = 0, RM = 1). In this operation mode ALIS will not decode a Caller ID.

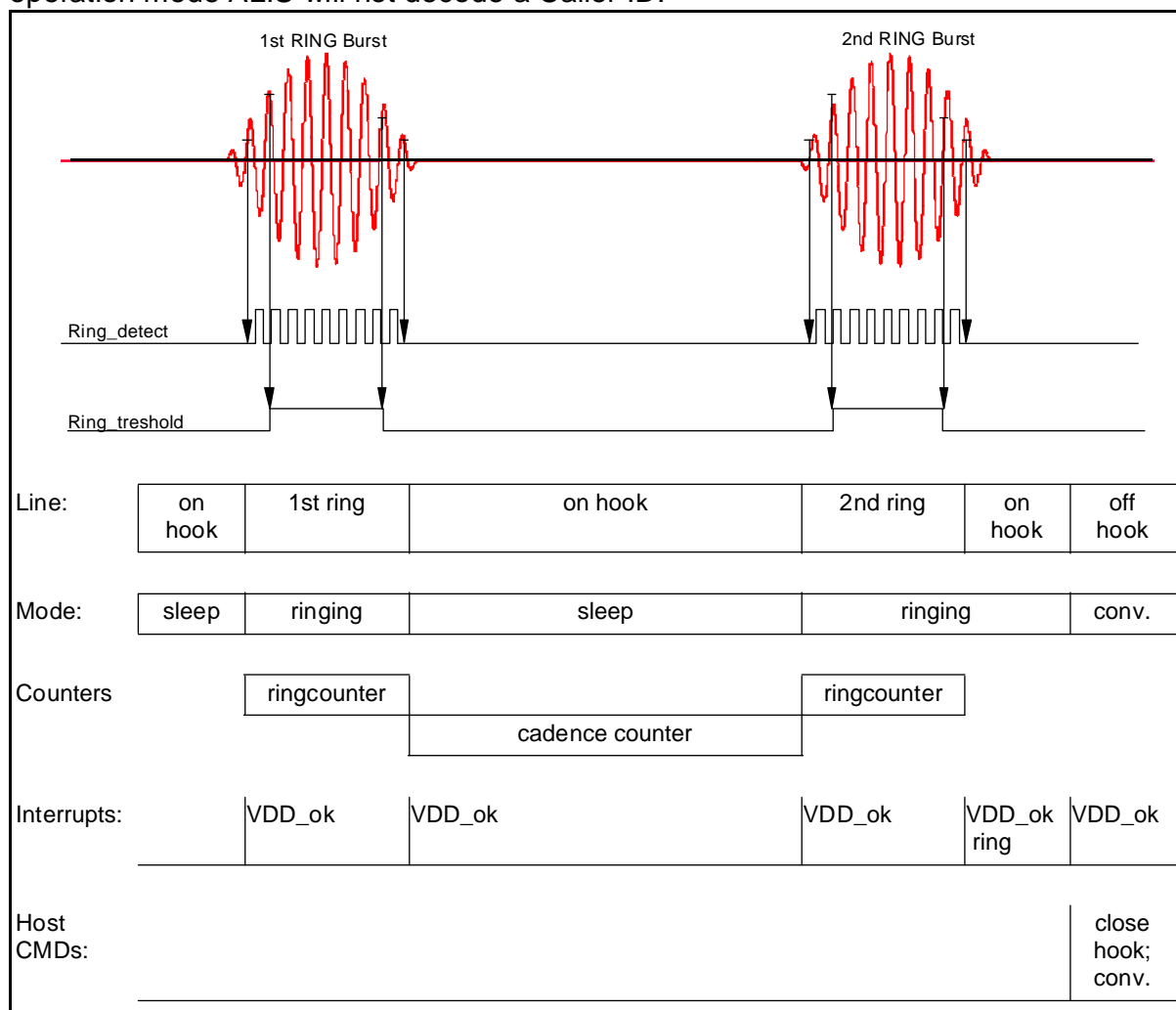


Figure 21: successful Ring sequence, No\_auto\_ring = 0; Caller\_en = 0



## 9.8.2 Successful Ring\_sequence, auto Ring enabled, Caller ID

The following chart and diagram shows a successful flow of a ring-event detection with automatic power mode change (No\_auto\_ring = 0, Caller\_en = 1, RM = 1). In this operation mode ALIS will decode and store a Caller ID.

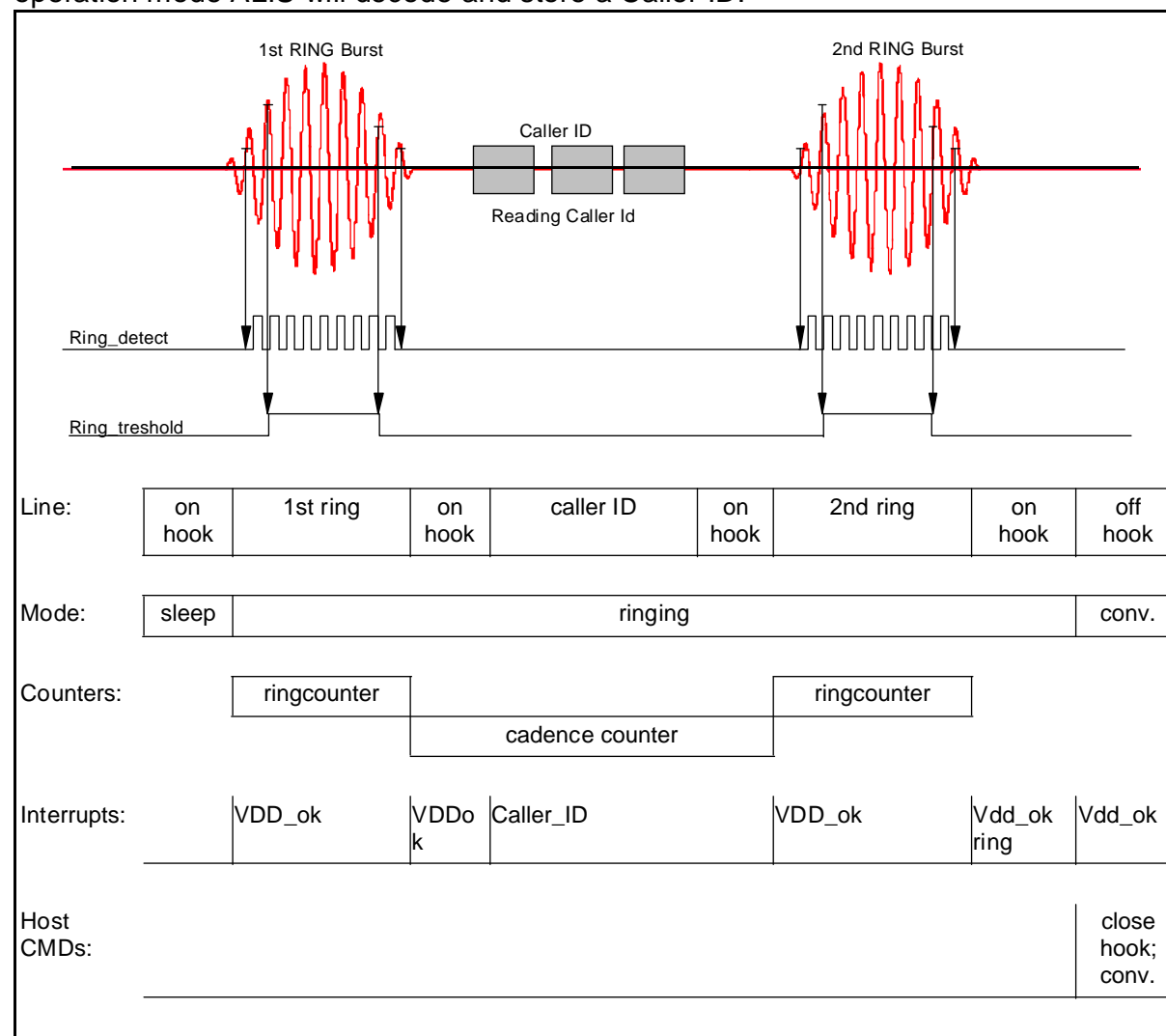


Figure 22: successful Ring sequence, No\_auto\_ring = 0; Caller\_en = 1

## 9.8.3 Not successful Ring\_sequence, auto Ring enabled, No Caller ID

The following chart and diagram shows a not successful flow of a ring-event detection because of no 2. ring with automatic power mode change (No\_auto\_ring = 0, Caller\_en = 0, RM = 1).

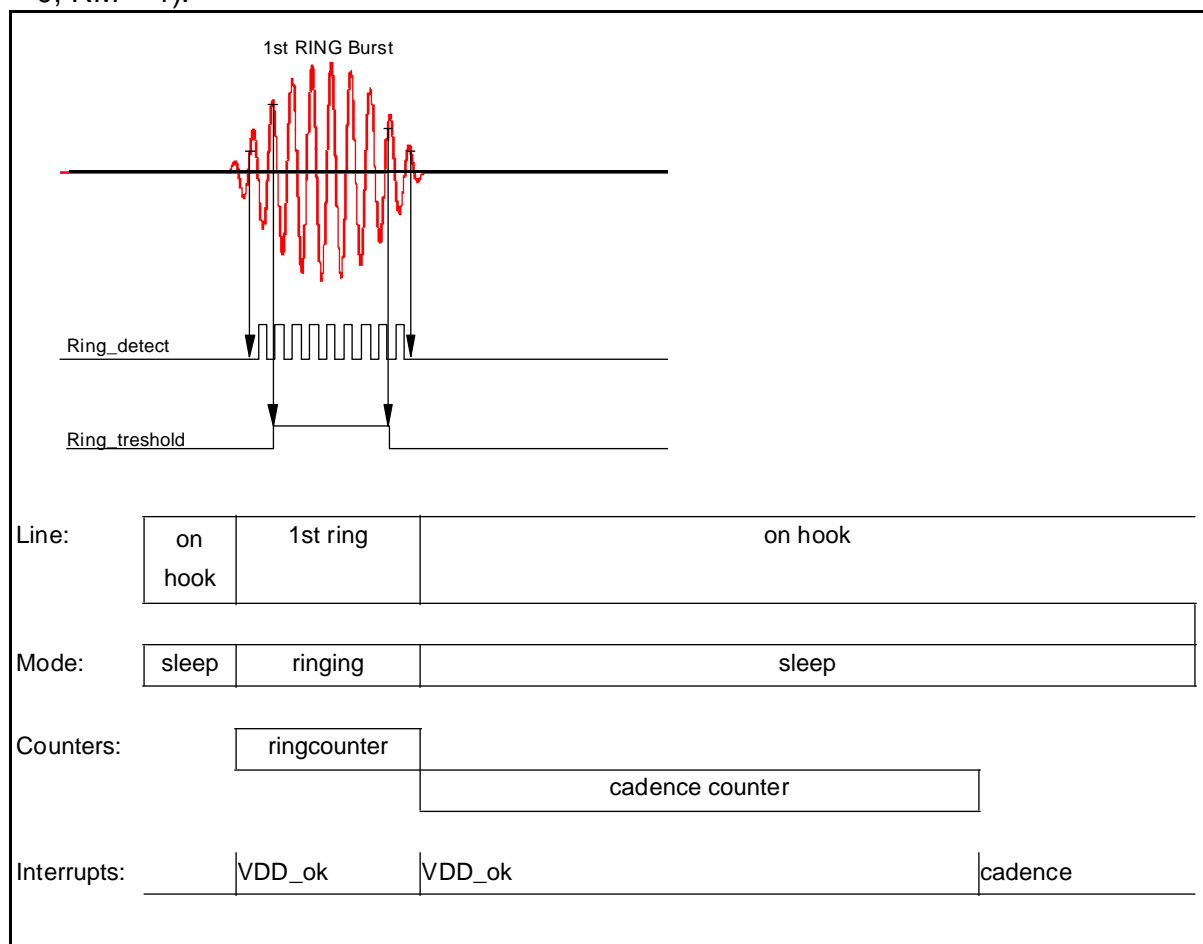


Figure 23: not successful Ring sequence, No\_auto\_ring = 0; Caller\_en = 0

## 9.8.4 Not successful Ring\_sequence, auto ring enabled, Caller ID

The following chart and diagram shows a not successful flow of a ring-event detection because of no 2. ring with automatic power mode change (No\_auto\_ring = 0, Caller\_en = 1, RM = 1).

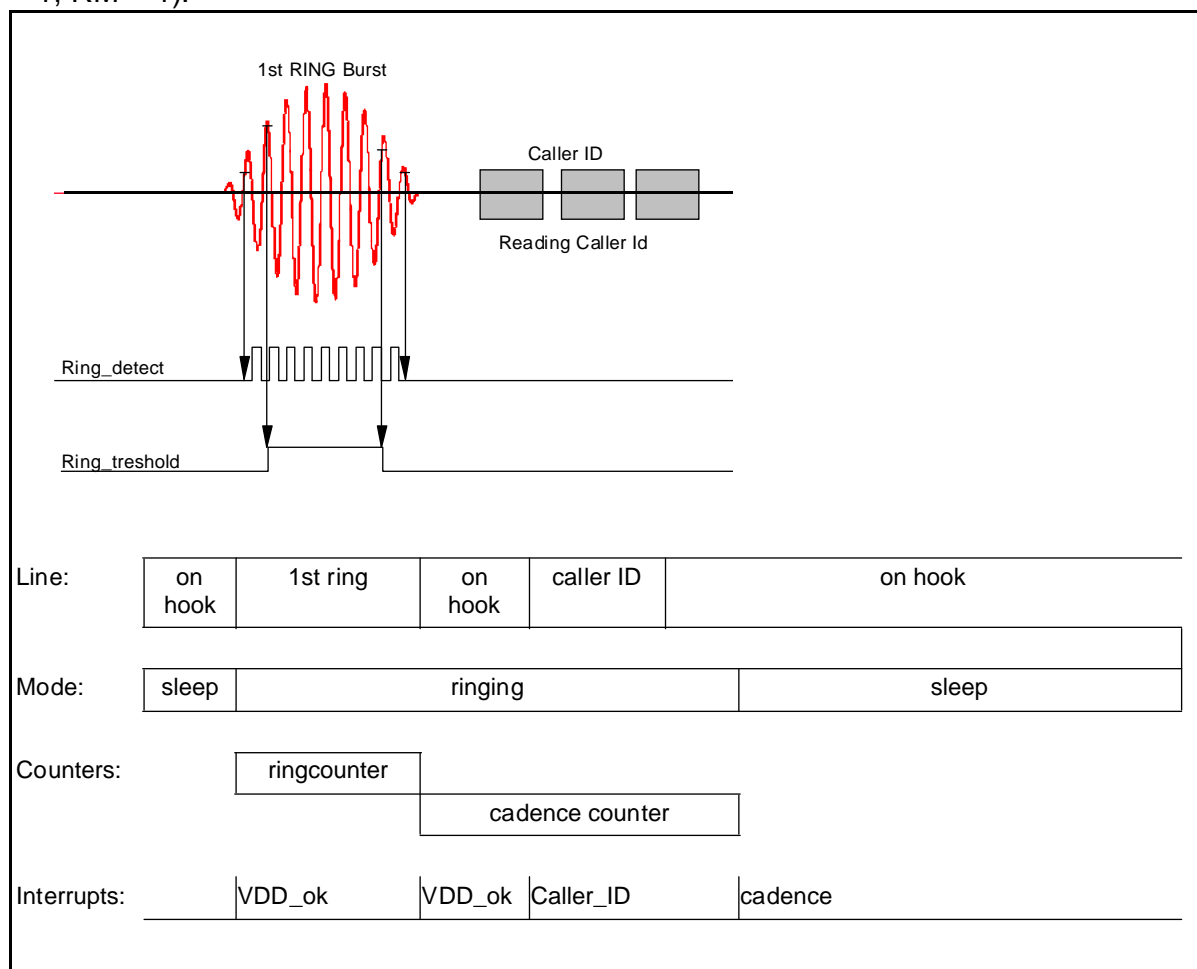


Figure 24: not successful Ring sequence, No\_auto\_ring = 0; Caller\_en = 1

## 9.8.5 Successful Ring\_sequence, auto Ring disabled, No Caller ID

The following chart and diagram show a successful flow of a ring-event detection with no automatic power mode change (No\_auto\_ring = 1, Caller\_en = 0, RM = 1). In this operation mode ALIS will not decode the Caller ID.

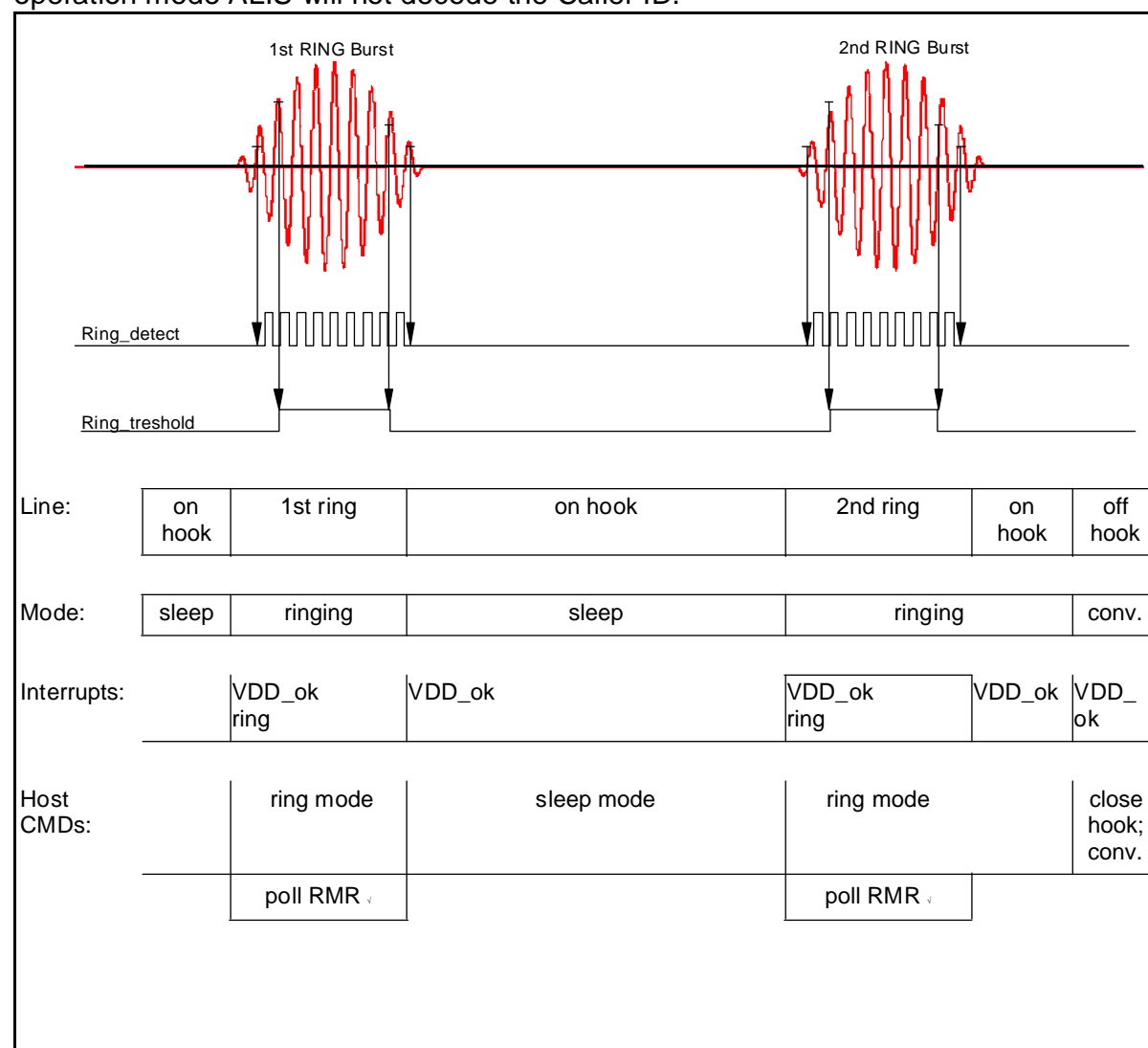


Figure 25: successful Ring sequence, No\_auto\_ring = 0; Caller\_en = 0

*Note: RMR bit must be polled by the host to verify that the ring signal is above the programmed threshold level and check VDD interrupts*

## 9.8.6 Successful Ring\_sequence, auto Ring disabled, Caller ID

The following chart and diagram show a successful flow of a ring-event detection with no automatic power mode change (No\_auto\_ring = 1, Caller\_en = 1, RM = 1). In this operation mode ALIS will decode and store the Caller ID.

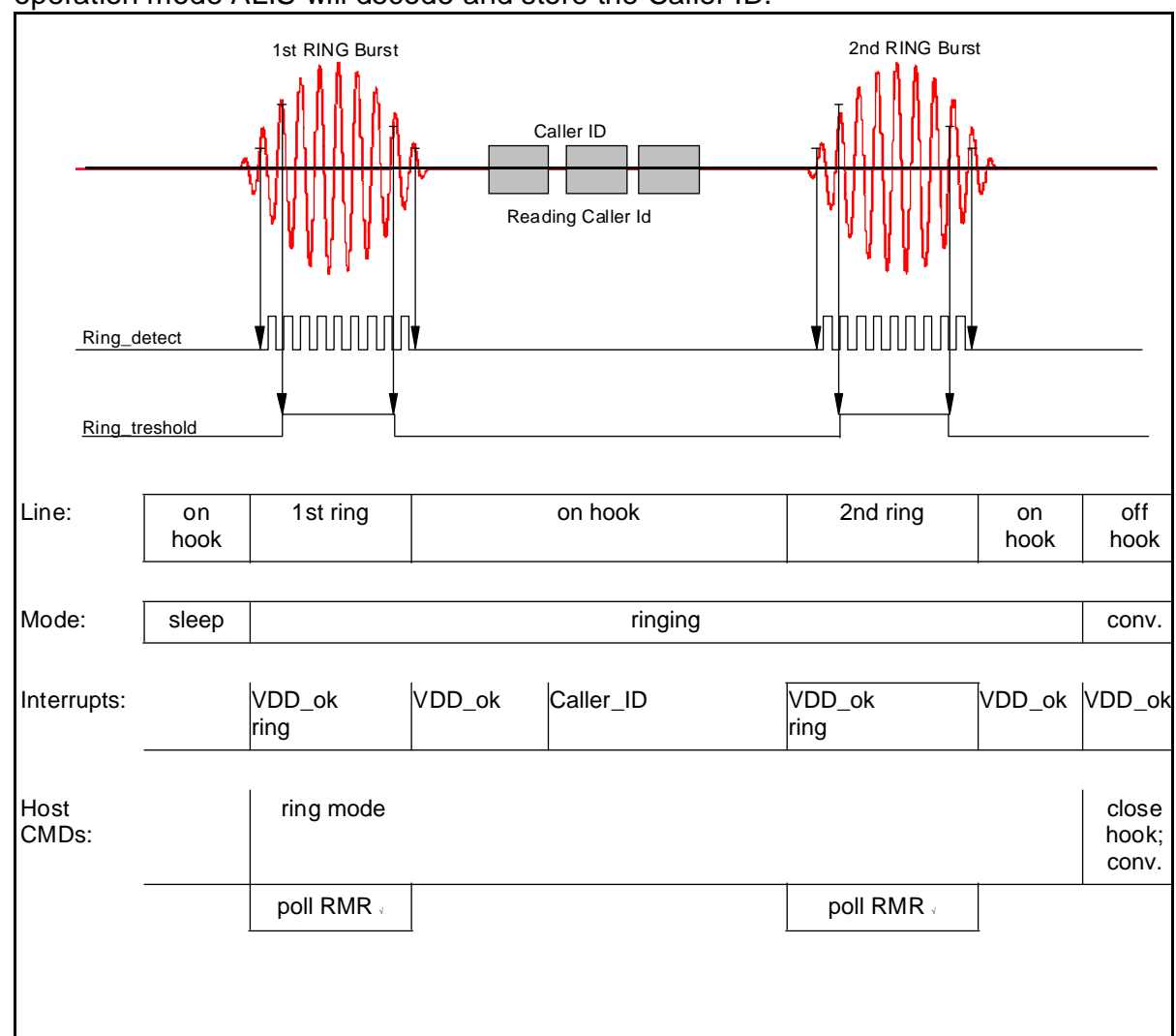


Figure 26: successful Ring sequence, No\_auto\_ring = 1; Caller\_en = 1

*Note: RMR bit must be polled by the host to verify that the ring signal is above the programmed threshold level.*

*Leaving ALIS in ringing mode after the first ring enables the detection and storage of the Caller ID.*

## 9.8.7 Not successful Ring\_sequence, auto Ring disabled, No Caller ID

The following chart and diagram show a not successful flow of a ring-event detection because of no 2. ring with no automatic power mode change (No\_auto\_ring = 1, Caller\_en = 0, RM = 1).

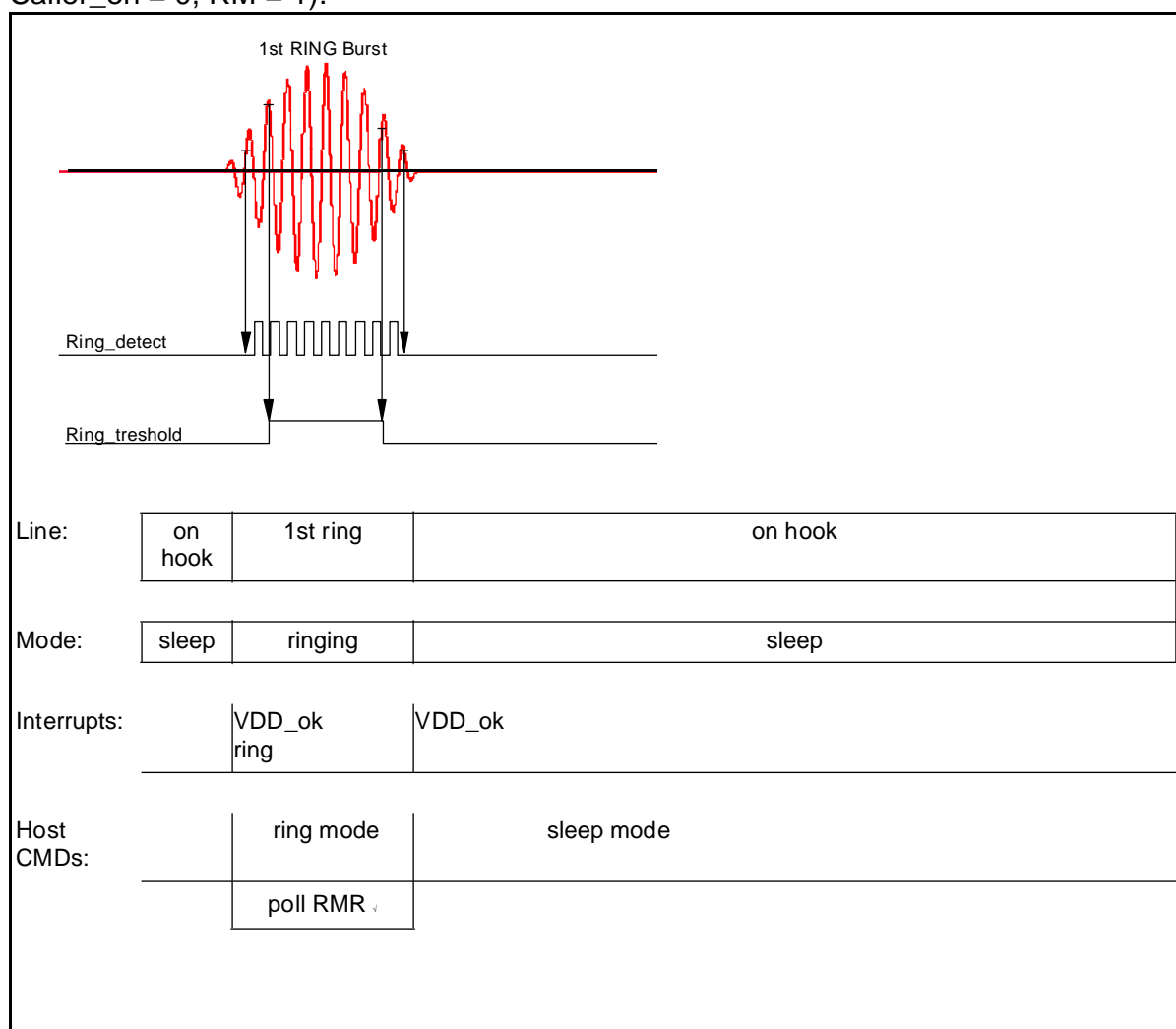


Figure 27: not successful Ring sequence, No\_auto\_ring = 1; Caller\_en = 0

*Note: RMR bit must be polled by the host to verify that the ring signal is above the programmed threshold level.*

*Cadence time and number of rings must be calculated by the host.*

## 9.8.8 Not successful Ring\_sequence, auto ring disabled, Caller ID

The following chart and diagram show a not successful flow of a ring-event detection because of no 2. ring with no automatic power mode change (No\_auto\_ring = 0, Caller\_en = 1, RM = 1).

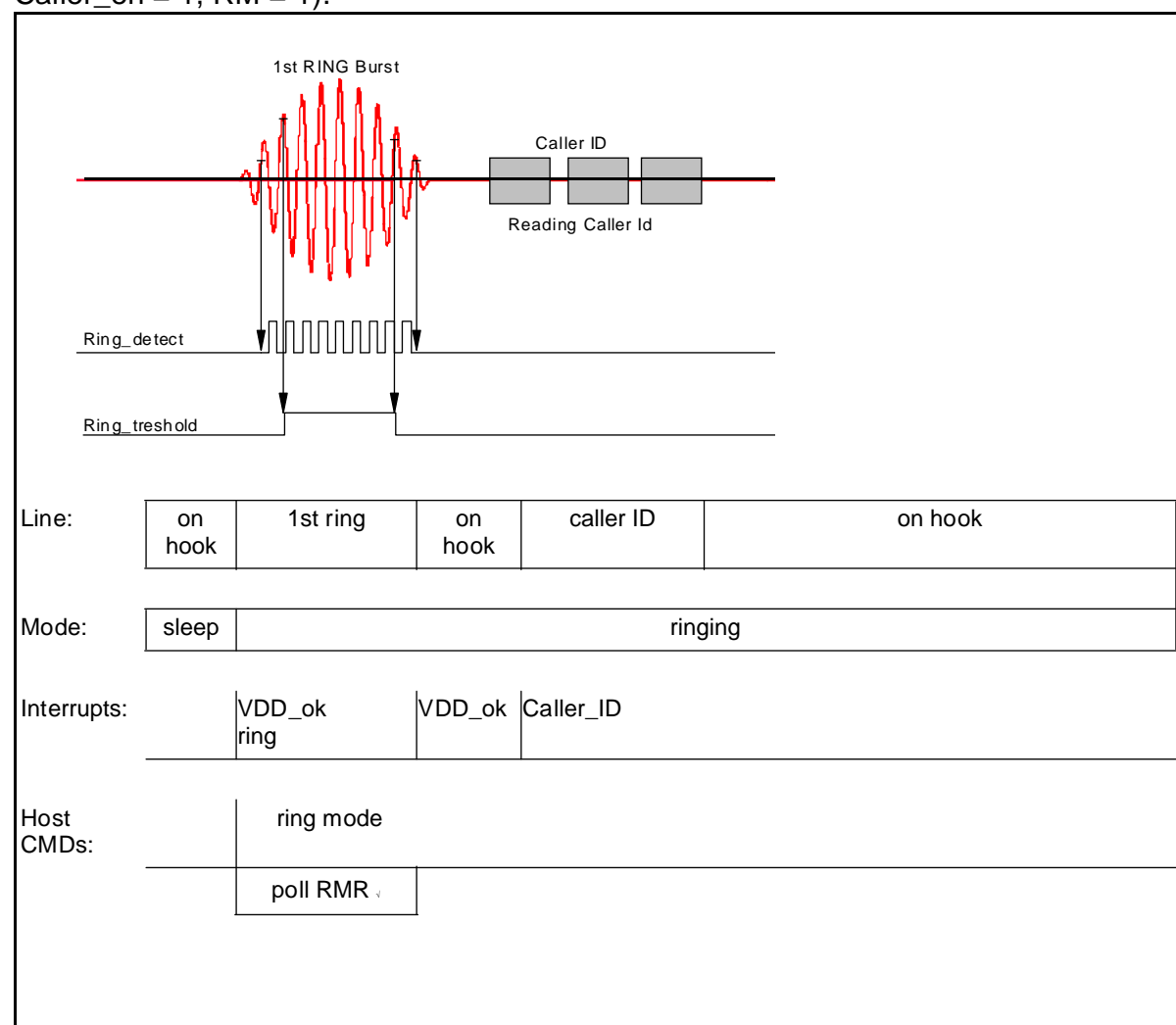


Figure 28: not successful Ring sequence, No\_auto\_ring = 1; Caller\_en = 1

*Note: Cadence time and number of rings must be calculated by the host.*

## 9.8.9 Not successful Ring\_sequence, auto Ring enabled

The following chart and diagram shows a not successful flow of a ring-event detection because below the ring threshold level ring with automatic power mode change (No\_auto\_ring = 0, RM = 1).

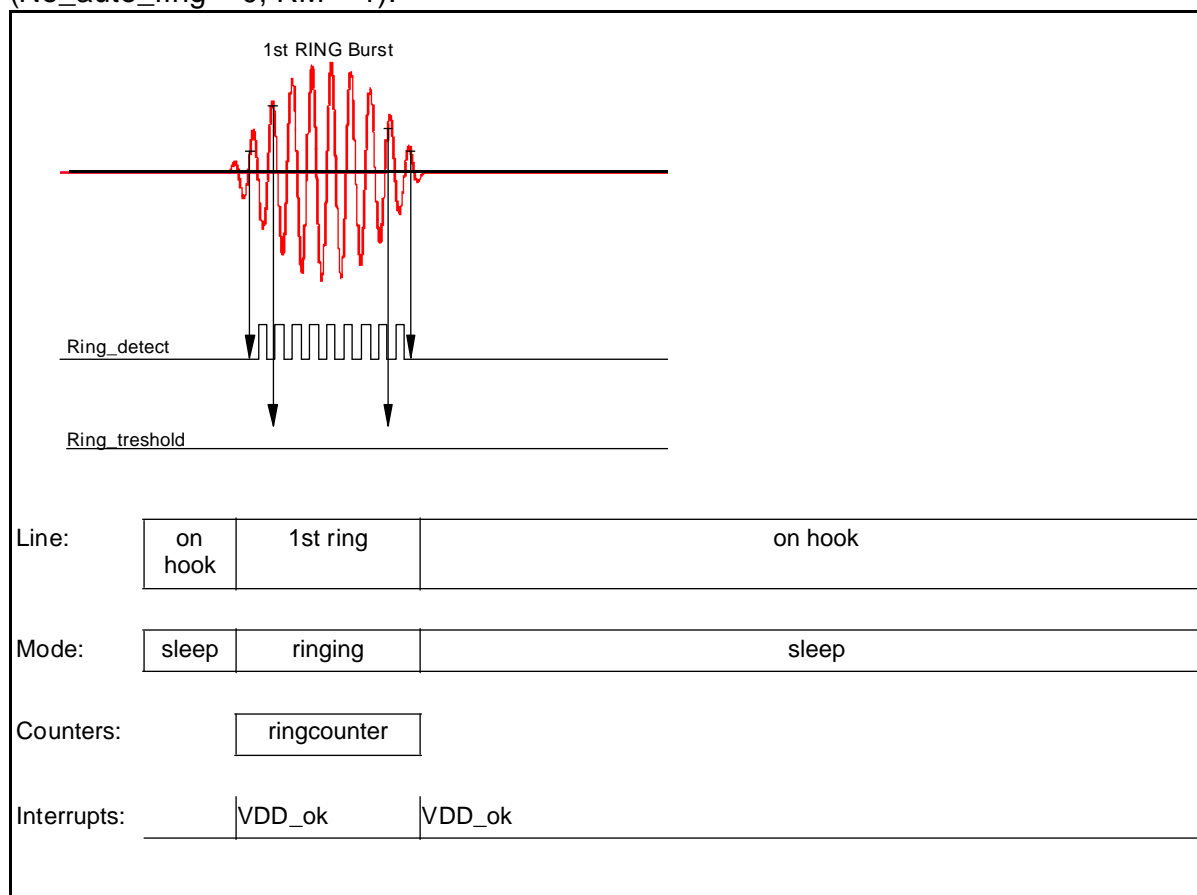


Figure 29: not successful Ring sequence, No\_auto\_ring = 0;



## 9.8.10 Not successful Ring\_sequence, auto Ring disabled

The following chart and diagram show a not successful flow of a ring-event detection because ringing is below the ring threshold level ring with no automatic power mode change (No\_auto\_ring = 1, RM = 1).

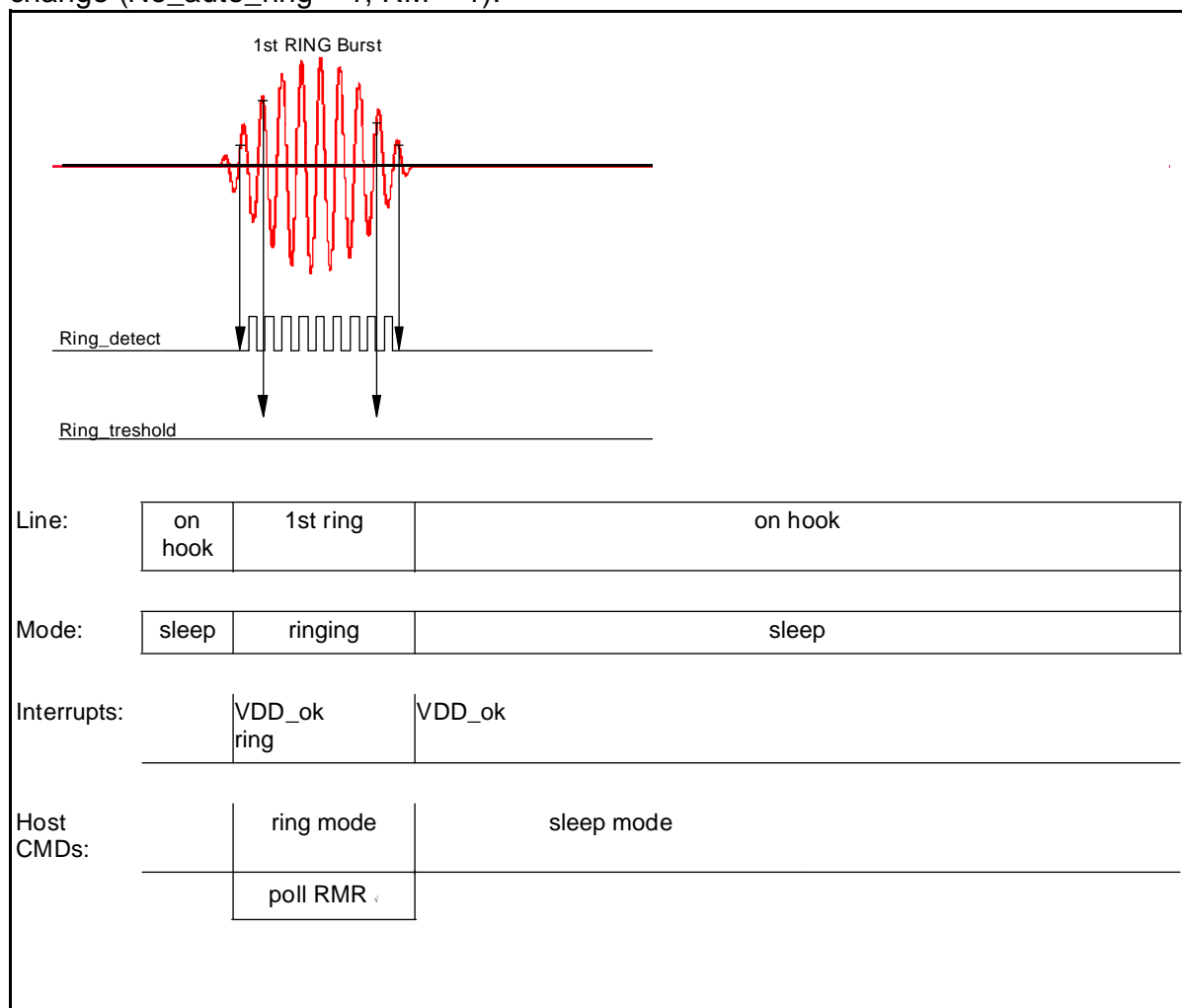


Figure 30: not successful Ring sequence, No\_auto\_ring = 0;

*Note: RMR will not be '1'*

## 9.8.11 Start from deep sleep mode

The following chart and diagram show a start-up procedure from deep sleep mode.

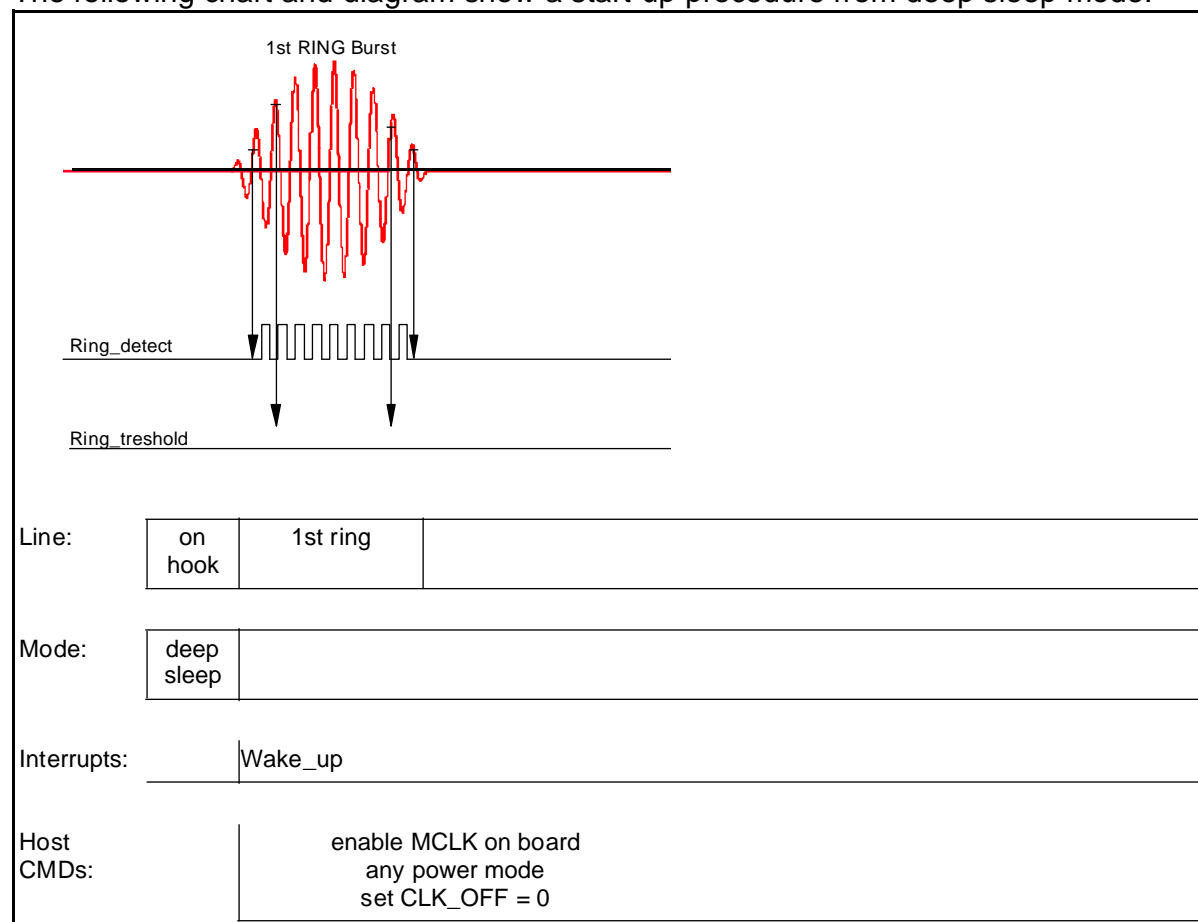


Figure 31: deep sleep start

*Note: after the wake\_up interrupt any power mode and operation flow according to the previous chapters can be programmed.*

## 10 Modem Functions

### 10.1 Pulse Dialing

Pulse dialing will be realized by shortening the line with the external transistor T1. Pulse timing has to be controlled by the host. Pulse shaping is realized in ALIS\_A, the pulse shaping will conform to ETS 300 001

### 10.2 DTMF Dialing

DTMF Dialing is realized by two internal tone generators (See "Programming the ALIS DTMF Tone Generators" on page 75). Since the level of tone generator 2 is 3 dB higher

than that of tone generator 1, it should be used for the high frequency group. The frequency accuracy of the tone generators is better than  $\pm 1\%$ . The absolute transmission level can be programmed using the AX filter. Software for computing the coefficients is available.

The tone generators can also be used to generate any inband sinewave for test or measurement purposes.

## 10.2.1 Programming the ALIS DTMF Tone Generators

Two independent tone generators are available. When one or both tone generators are turned on the voice signal is switched off automatically. To make the generated signal suitable for DTMF a programmable bandpass-filter is included. The default frequency for both tone generators is 2000Hz. Coefficients for other frequencies are generated by a software tool.

Byte sequences for programming both tone generators and the bandpass filters:

Frequency	Command	Byte 1	Byte 2	Byte 3	Byte 4
697 Hz	0B *)	11	B3	5A	2C
770 Hz	0B *)	12	33	5A	C3
852 Hz	0B *)	13	3C	5B	32
941 Hz	0B *)	1D	1B	5C	CC
1209 Hz	0C *)	32	32	52	B3
1336 Hz	0C *)	EC	1D	52	22
1477 Hz	0C *)	AA	AC	51	D2
1633 Hz	0C *)	9B	3B	51	25

\*) 0B is used for programming Tone Generator 1  
0C is used for programming Tone Generator 2

Table 7: Programming the Tone generators

The sinewave will be filtered by a bandpass, the Q factor of this bandfilter can be altered in the range from 0 to 7 and can be programmed by setting the first nibble of byte 3 to the corresponding value (always 5 in this table). The resulting signal amplitude can be set by programming the AR1 and AR2 filters.

## 10.3 Caller ID

The Caller ID interface is compatible to the Bellcore TR-NWT-000030 and SR-TSV-002476 on generic requirements for transmitting asynchronous voiceband data to Customers Premises Equipment (CPE) from a serving Stored Control Switching System (SPCS) or a Central Office (CO). In this service, the information about the calling party is embedded into the silent interval between the first and the second ring. During this period ALIS receives and stores up to 4096 bits of the 1200 baud FSK signal. The decoding matches as well BELL 202 and CCITT V.23 specifications. (See "Programming the ALIS Caller ID coefficients" on page 77)

### 10.3.1 Characteristics for Caller ID

Parameter	Symbol	Limit Values			Unit	Reference
		min	typ	max		
Input detection level	V <sub>in</sub>	-36 12.3		-9 275	dBm mV	
Detect frequencies						
Bell 202 1 (Mark)		1188	1200	1212	Hz	Bell 202
Bell 202 0 (Space)		2178	2200	2222	Hz	
CCITT V.23 1 (Mark)		1280.5	1300	1319.5	Hz	CCITT V.23
CCITT V.23 0 (Space)		2068.5	2100	2131.5	Hz	
Input Noise Tolerance	SNR	20			dB	
Input Baud Rate		1188	1200	1212	Hz	

Table 8: Characteristics for Caller ID

### 10.3.2 Storage and reading of Caller ID

The storage of the decoded Caller ID is enabled after the first space following the mark state. This event will be indicated by the Caller ID interrupt. The maximum storage size is 4096 bits. Start, stopbit and checksum decoding must be done by the host.

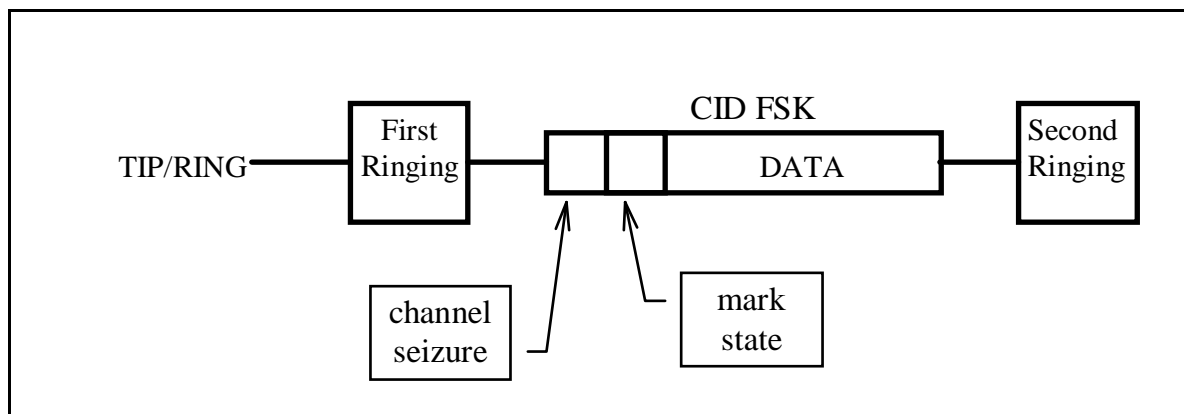
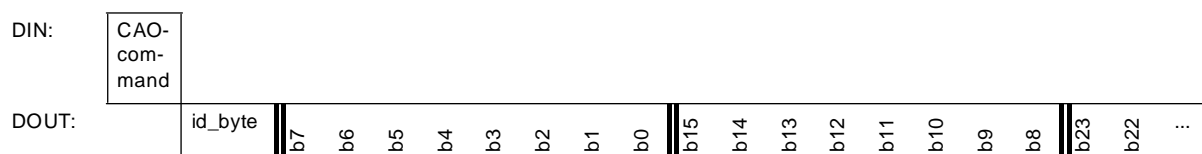


Figure 32: CID Input timing

When reading the ram with the CAO command the received bits will be sent from ALIS in the following order:



b0 is the first caller- ID data bit after the '0' which is ending the marker sequence, b1 the second, b2 the third etc. ...

The host can read the caller- ID RAM at any time. Note that the read data may be erroneous when caller-ID data is received at the same time as old and new data might be mixed. However the received caller-id bits are stored correctly in the ram!

-> Try not to read the ram during caller-id is received!

### 10.3.3 Programming the ALIS Caller ID coefficients

Frequency	Command	Byte 1,2	Byte 3,4	Byte 5,6	Byte 7,8
BELL 202 / CCITT V.23	0E (CID1)	CA 0E	CA 09	99 99	99 99
	0F (CID2)	FD B5	BA 07	DA XX	XX XX

Table 9: Programming the ALIS Caller ID coefficients

## 10.4 Billing pulse

Billing pulse frequencies of 12 and 16 kHz will be filtered out by the digital part of ALIS, there are no external components for blocking necessary.

## 10.5 Ring Detect

### 10.5.1 Functional description

In Sleep Mode any signal greater then typical 18 volts will be detected. Depending on the No\_auto\_ring bit either an interrupt will occur or ALIS will be switched automatically into the Ringing Mode. In this mode the Ring Signal will be handed over to ALIS\_D and decoded. If the Ring Burst does not meet the programmed requirements within a programmable time ALIS will return to the Sleep Mode. After a latency time ALIS will decode the Caller ID. When the second valid Ring Burst occurs a Ring interrupt is generated, signalling the incoming call to the host (see See “Flow of Ring sequence and detection” on page 62)

### 10.5.2 Programming the ALIS Ring Detect Coefficients

Frequency	Command	Byte 1	Byte 2	Byte 3	Byte 4
25 Hz 70Vrms 10 kΩ 1.0 μF	0D	AA	05	0F	8E
	<b>Command</b>	<b>Byte 1,2</b>	<b>Byte 3,4</b>	<b>Byte 5,6</b>	<b>Byte 7,8</b>
	03	1C B3	AB AB	54 2D	62 2D
	06	2D 62	A6 BB	2A 7D	0A D4

Table 10: Programming ALIS Ring Detect coefficients

Frequency	Command	Byte 1	Byte 2	Byte 3	Byte 4
50 Hz 50Vrms 10kΩ 0.6 μF	0D	22	15	B5	84
	<b>Command</b>	<b>Byte 1,2</b>	<b>Byte 3,4</b>	<b>Byte 5,6</b>	<b>Byte 7,8</b>
	03	1C A4	AA AB	BD 2B	A2 2D
	06	2B A2	A6 BB	2C 63	3A D4

Table 11: Programming ALIS Ring Detect coefficients

## 10.5.3 Ring Threshold in Sleep Mode

Parameter	Symbol	Limit Values			Unit	Reference
		min	typ	max		
Ring Threshold			12	18	Vrms	

Table 12: Ring Threshold in Sleep Mode

## 11 Electrical Characteristics

### 11.1 Programmable Filters

A set of programmable filters are used to adapt the whole system to:

- country standards
- board designs (EMI capacitors etc.)
- data pumps
- telephone lines

*Note: All this coefficients will be computed by a coefficient program. Any change of these computed values may case a loss of performance or instability.*

In detail the following filters are programmable:

- Transhybrid Balancing (TH)-Filter
- Transhybrid Prebalancing
- Impedance Matching (IM)-Filter
- Frequency Response Receive (FRR)-Filter
- Frequency Response Transmit (FRX)-Filter
- Ringer Impedance

Amplification/Attenuation Transmit (AX)-Filter

Gain for AX-Filter

range 3.. -14 dB: step size 0.02 .. 0.05 dB  
range -14 .. -24 dB: step size 0.5 dB

Gain for AGX

range 3.5, 0 -2.5, -6

Amplification/Attenuation Receive (AR)-Filter

Gain for AR-Filter

range -3 .. 14 dB: step size 0.02 .. 0.05 dB  
range 14 .. 24 dB: step size 0.5 dB

Gain for AGR\_R:

range 0, 3.5, 6 dB

Gain for AGR\_Z:

range -3.5, 0, 2.5, 6 dB

## 11.2 DC Characteristics

The Filtercoefficients are generated by a softwaretool including a high level model of ALIS and additional, user defined or application specific system components.

### 11.2.1 DC Termination

The DC termination is enabled in Conversation mode and disabled during Ringing Mode, Pulse Dialing Mode and Sleep Mode. The DC Termination can be programmed according the formula:

for  $i < I_{max}$

$$i(u) = \frac{(u - U_0)}{R}$$

for  $i > I_{max}$

$$i(u) = I_{max}$$

*Note:*  $U_0$  is the sum of  $U$  listed in table 14 and the flow voltage of the diodes in the external bridge (typ.  $2 \times 0.4$  V)

### 11.2.2 Programming ranges for DC Termination

<b>I<sub>max</sub></b>
50 mA
100 mA

Table 13: Programming range for I<sub>max</sub>

<b>U (DCU)</b>
0 V
1.5 V
3.5 V
7.2 V

Table 14: Programming range for U

<b>R (DCR)</b>
70 Ω



100 $\Omega$
200 $\Omega$
240 $\Omega$
280 $\Omega$

Table 15: Programming range for R

*Note: for programming details see See "XR3 Extended Register 3 (DC Characteristic)" on page 46*

### 11.2.3 Input Current in Pulse Dialing Mode

$U_{ab} = 30 \text{ V DC}$

Parameter	Symbol	Limit Values			Unit
		min	typ	max	
Input current at break	lin			500	$\mu\text{A}$

Table 16: Input current in pulse dialing

## 11.3 AC Termination

### 11.3.1 Ringer Impedance

The programming of the Ringer Impedance is supported by a software tool and the following table shows typical values.

$U_{ab} = 70 \text{ Vrms}$

Parameter	Symbol	Limit Values			Unit
		min	typ	max	
Ringer Impedance (20 Hz < f < 60 Hz) <sup>1)</sup>	Rin		10		$\text{k}\Omega$
	$\Delta \text{ Rin}$			tbd	%
Typical capacitors	Cin1		0.6		$\mu\text{F}$
	Cin2		1		$\mu\text{F}$
	$\Delta \text{ Cin}$			tbd	%
Ringer Impedance in other modes <sup>2)</sup>					

- 1) The frequency range can be changed
- 2) Ringer impedance is generated only in ring mode

Table 17: Ringer Impedance

## 11.4 ALIS Caller ID Interface

Parameter	Symbol	Limit Values			Unit
		min	typ	max	
Capacitance	Cin		50		nF
Tolerance between Cin1 and Cin2				tbd	%
Rin	Rin		50		kΩ
Tolerance between Rin1 and Rin2				tbd	%
Rfb	Rfb		200		kΩ
Tolerance between Rfb1 and Rfb2				tbd	%

Table 18: ALIS Caller ID interface

### 11.4.1 Ring Detect Levels and Frequencies

Parameter	Symbol	Limit Values			Unit	Tolerance
		min	typ	max		
Ring level detection program range	Vring	30		100	V	±10%
Ring level detection step size	Δ Vring			10	V	±10%
Ring Frequency detection program range <sup>1)</sup>	Fring	20		60	Hz	±10%

- 1) The exact range depends on the level programmed for detection

Table 19: Ring detect levels and frequencies

**Electrical Performance Characteristic**
**11.5 ALIS Cap Interface**

Parameter	Symbol	Limit Values			Unit
		min	typ	max	
Capacitance	Cin	5	10	30	pF
Tolerance between CAP_x1 and CAP_x2				5	%
Inductance				10	nH
Isolation	tbd.				kV

Table 20: ALIS Cap interface

**12 Electrical Performance Characteristic**
**12.1 Absolute Maximum Ratings**

Parameter	Symbol	Ratings		Unit
		Min	Max	
Digital Supply Voltage	VDD	-0.3	7.0	V
Analog Supply Voltage	VDDA	-0.3	7.0	V
Analog Input and Output Voltage	Vin, Vout	-0.3	VDDA + 0.3	V
Digital Input Voltages	VDin	-0.3	VDD + 0.3	V
DC Input and Output Current	Iin, Iout	-10	10	mA
Storage Temperature	TST	-60	125	°C
Ambient Temperature under Bias	TA	-10	80	°C
Max. Power Dissipation	PDmax		1	W

*Note: Stresses above the absolute maximum ratings may cause permanent damage to the device. Extended operation at the maximum levels may degrade performance and effect reliability.*

Table 21: Absolute Maximum Ratings

**Electrical Performance Characteristic**
**12.2 Recommended Operating Conditions**

Parameter	Symbol	Conditions			Unit
		Min	Typ	Max	
Digital Supply Voltage	VDD	4.75	5.0	5.25	V
Analog Supply Voltage ALIS_A (programmed to 4.25 V)	VDDA	4.00	4.25		V
Analog Supply Voltage ALIS_D	VDDA	4.75	5.0	5.25	V
Ambient Temperature under Bias	TA	0		70	°C
Operating Frequency	fclk		16.384	tbd.	MHz
Clock Duty Cycle		45	50	55	%
Signal Rise and Fall Time	tr, tf			20	ns

*Note: Extended operation outside the recommended limits may degrade performance and effect reliability.*

Table 22: Recommended Operating Conditions

**12.3 DC Characteristics**
**12.3.1 ALIS\_A**

VDDA= 4.25V progr.; TA=0 - 70°C

Parameter	Symbol	Conditions	Spec. Limits			Unit
			Min	Typ	Max	
Power-Up Time	tPU	tbd			tbd	ms
VDDA Supply Current <sup>1)</sup>						
Ringing Mode <sup>2)</sup>	IDDA1	Vring=60V DC + 90Vrms, fring=25 - 50Hz		2.5	tbd	mA
Conversation Mode <sup>3)</sup>	IDDA2	fCLK=16.384MHz		7	tbd	mA
Pulse Dialing Mode	IDDA3	Vab=30 V DC			500	µA
Digital Interface						
Low-Level Input Voltage	VIL <sup>4)</sup>				0.8	V

## Electrical Performance Characteristic

High-Level Input Voltage	VIH <sup>4)</sup>		2.0			V
Low-Level Output Voltage	VOL <sup>5)</sup>	IOL=5mA			0.5	V
High-Level Output Voltage	VOH <sup>5)</sup>	IOH=-5mA	3.5			V
Input Current Low	IIL	VIL=GNDA			±1	μA
Input Current High	IIH	VIH=VDDA			±1	μA
Input Resistance						
Sleep Mode	Rin	@ 100 V DC	1			MΩ
Conversation Mode	Rin	see 10.2.3				Ω
Pulse Dialing Mode	Rin	Interpulsing Period (Make)			200	Ω
Ring Threshold	VRThresh	VDDA=4.25V ext.			15	VRMS
Power Supply Rejection	PSRR	Ripple: 0-150kHz; 70mVrms				
either supply/direction		300Hz - 3.4kHz	tbd			dB
VDD receive guaranteed		3.4kHz - 150kHz	tbd			dB
VDD receive Target		3.4kHz - 150kHz	tbd			dB

- 1) Will be taken from TIP/RING when the hook switch is open
- 2) in Ringing mode the ringer impedance will be synthesized. Therefore a current according to this impedance will flow from TIP/RING. This current is taken out of the ring burst as an ac current.
- 3) in Conversation mode the DC characteristic will be synthesized and a current according to this characteristic will flow from TIP/RING.
- 4) Digital Inputs: Test, SI\_0, SI\_1
- 5) Digital Outputs: SO\_0, SO\_1Q

Table 23: DC Characteristics ALIS\_A

**Electrical Performance Characteristic**
**12.3.2 ALIS\_D**

VDD = VDDA = 5V ± 5%; TA = 0 - 70°C

Parameter	Symbol	Conditions	Spec. Limits			Unit
			Min	Typ	Max	
Supply Current		VDD=5V, no loads				
Deep Sleep Mode	IDD0				tbd	μA
Sleep Mode	IDD1			1	1	mA
Ringing Mode	IDD2			3	3	mA
Conversation Mode	IDD3			10	15	mA
Pulse Dialing Mode	IDD4			10	15	mA
Low-Level Input Voltage	VIL <sup>1)</sup> VIL <sup>2)</sup> VIL <sup>3)</sup>				0.8 1.5 1.5	V
High-Level Input Voltage	VIH <sup>1)</sup> VIH <sup>2)</sup> VIH <sup>3)</sup>		2.0 3.5 3.5			V
Low-Level Output Voltage	VOL <sup>4)</sup>	IOL=5mA			0.5	V
High-Level Output Voltage	VOH <sup>4)</sup>	IOH=-5mA	VDD-0.5			V
Input Current Low	IIL <sup>1,2)</sup>	VIL=GND			±1	μA
Input Current High	IIH <sup>1,2)</sup>	VIH=VDD			±1	μA
Tristate Current Low	IOZL <sup>5)</sup>	VIL=GND			±1	μA
Tristate Current High	IOZH <sup>5)</sup>	VIH=VDD			±1	μA

1) TTL Inputs: DCLK, CS, DIN, DAT\_CLK, DAT\_IN, MODE, FSC

2) CMOS Input: RESET

3) Clock Input: MCLK1

4) Outputs: DOUT, INT, DAT\_OUT, FSC

5) Tristates, Bidirectionals: DOUT, FSC

Table 24: DC Characteristics ALIS\_D

## 12.4 AC Transmission Characteristics

Unless otherwise stated, the transmission characteristics are guaranteed within the following test conditions:

TA=0 °C to 70 °C

VDD=5V ±5%

VDDA=4.25V (generated from ALIS\_A)

Line impedance ZL = 600 ± 0.1% Ohms

Termination impedance ZM = 600 Ohms

digital: 0dBm0 = -3 dB FS

analog: 0 dBm is equal to the voltage of 0.775 Vrms when loaded with 600 Ohms

0 dBm = 0dBm0

f=1004Hz.

### 12.4.1 Absolute Gain Error

AGX=AGR=0 dB

Parameter	Symbol	Limit Values			Unit	Test condition
		min	typ	max		
Absolute gain error receive	AE_R					-10 dBm
TA=25 °C; VDDA=4.25V		-1	±0.5	+1	dB	
TA=0-70 °C; VDDA=4.25V		-1.2	±0.7	+1.2	dB	
Absolute gain error transmit	AE_X					-10 dBm0
TA=25 °C; VDDA=4.25V		-1	±0.5	+1	dB	
TA=0-70 °C; VDDA=4.25V		-1.2	±0.7	+1.2	dB	

Table 25: Absolute Gain Error

## Electrical Performance Characteristic

### 12.4.2 Gain Tracking

AGX=AGR=0dB

Parameter	Symbol	Limit Values			Unit	Test condition
		min	typ	max		
Gain tracking receive	GT_R	-0.15		0.15		0 to -10 dBm
		-0.15		0.15		-30 to -40 dBm
		-0.3		0.3		-40 to -50 0.dBm
Gain tracking transmit	GT_X	-0.5		0.5		0 to -10 dBm0
		-0.1		0.1		-10 to -40 dBm0
		-0.5		0.5		-40 to -50 dBm0

Table 26: Gain Tracking

### 12.4.3 Harmonic Distortion plus Noise

-10 dBm0; ZL= 600 Ω; f=1004 Hz

Parameter	Symbol	Limit Values			Unit	Test condition
		min	typ	max		
HDN receive	THD+N_R	tbd	tbd		dB	C-weighted
HDN transmit	THD+N_T	tbd	tbd		dB	
HDN receive		tbd	tbd		dB	linear-weighted
HDN transmit		tbd	tbd		dB	
HDN of echo signals via TIP/RING		tbd	tbd		dB	C-weighted <sup>1)</sup>

1) measured with digital loop back via TIP/RING

Table 27: Harmonic Distortion plus Noise

### 12.4.4 Harmonic Distortion

-10 dBm0; ZL= 600 Ω; f=1004 Hz, 2nd and 3rd Harmonic

Parameter	Symbol	Limit Values			Unit	Test condition
		min	typ	max		
HD of echo signals via TIP/RING		80			dB	linear-weighted <sup>1)</sup>



1) measured with digital loop back via TIP/RING

Table 28: Harmonic Distortion

### 12.4.5 Return Loss

The return loss at a level of 0 dBm0, will be better than 16 dB in a 300-3600 Hz bandwidth using the following set of defined impedances

600 Ohms

220 Ohms + (820 Ohms in parallel with 115 nF)

120 Ohms + (820 Ohms in parallel with 110 nF)

370 Ohms + (620 Ohms in parallel with 310 nF)

### 12.4.6 Frequency Response

#### 12.4.6.1 Receive

Reference frequency 1kHz, input signal level 0dBm0

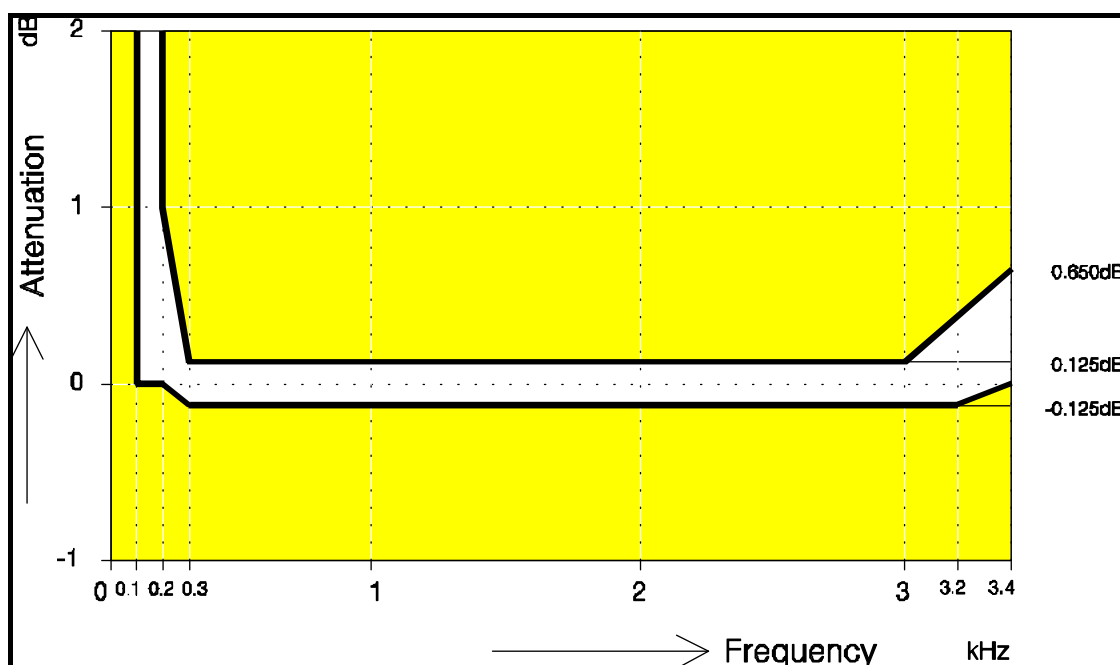


Figure 33: Frequency Response Receive

### 12.4.6.2 Transmit

Reference frequency 1kHz, input signal level 0dBm0

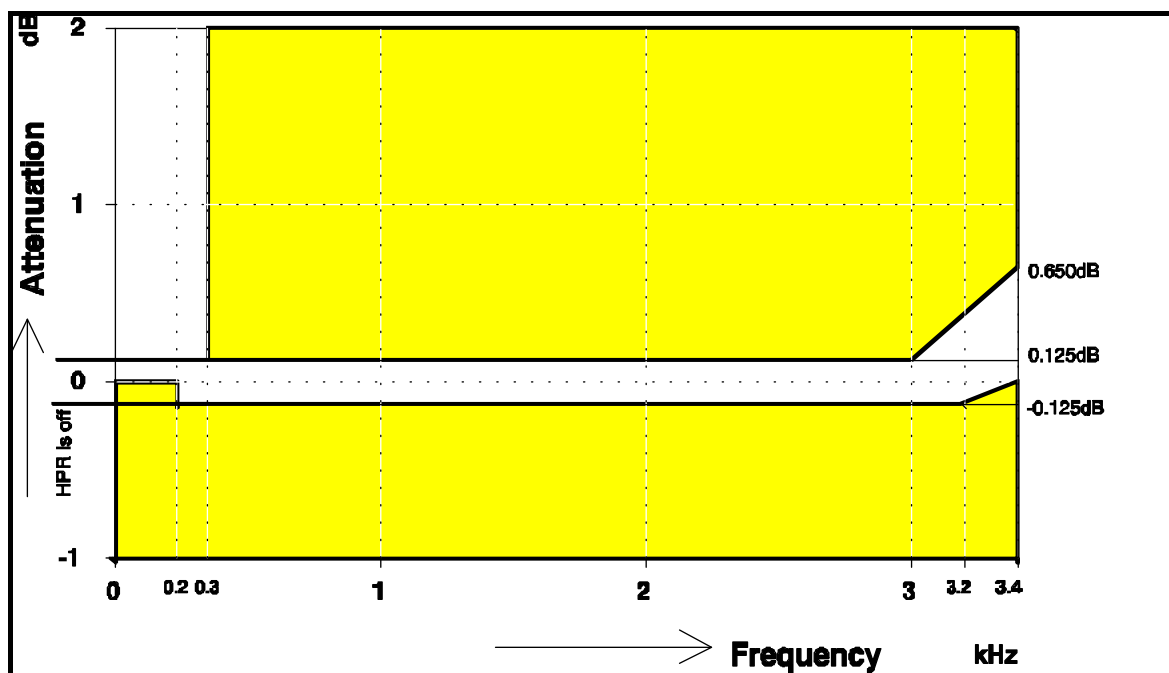


Figure 34: Frequency Response Transmit

### 12.4.7 Group Delay

Maximum delays when ALIS is operating with  $H(TH)=H(IM)=0$  and  $H(FRR)=H(FRX)=1$  including delay through A/D- and D/A converters. Specific filter programming may cause additional group delays.

Group Delay deviations stay within the limits in the figures below.

#### 12.4.7.1 Group Delay Absolute Values

Parameter	Symbol	Limit Values			Unit	Reference
		min	typ	max		
Receive delay	DRA			340	$\mu s$	Input signal level 0 dBm0
Transmit delay	DXA			400	$\mu s$	

Table 29: Group Delay

### 12.4.7.2 Group Delay Distortion Receive

Input signal level 0dBm0

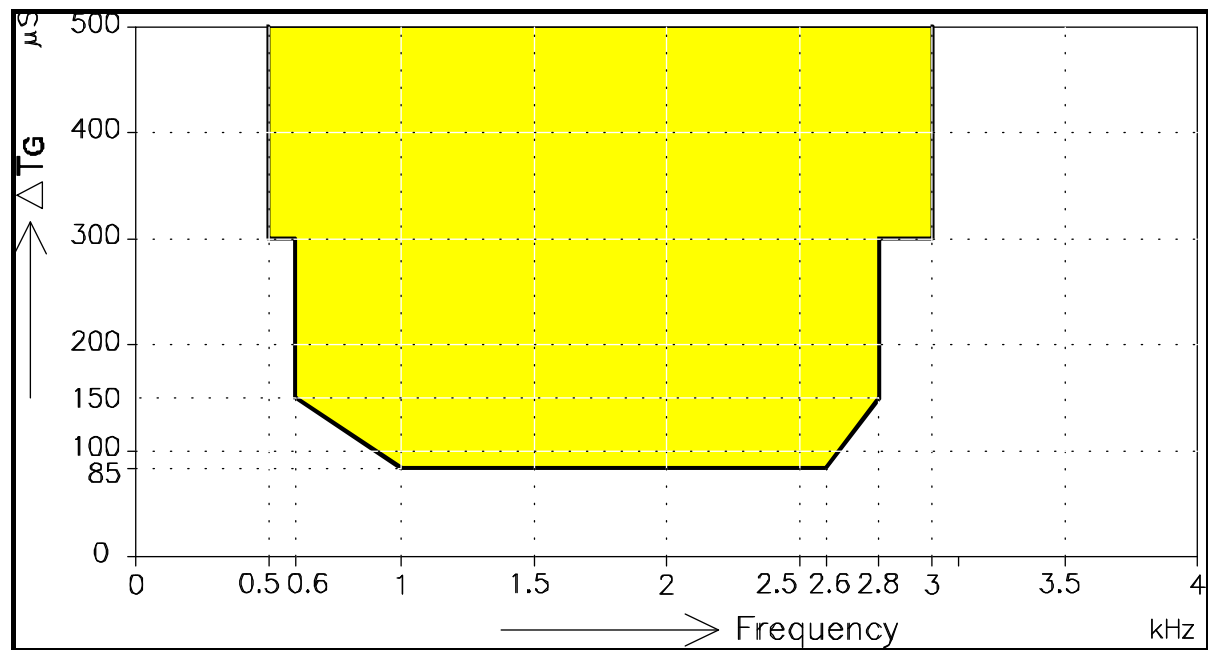


Figure 35: Group Delay Distortion Receive

## 12.4.7.3 Group Delay Distortion Transmit

Input signal level 0dBm0 <sup>1)</sup>

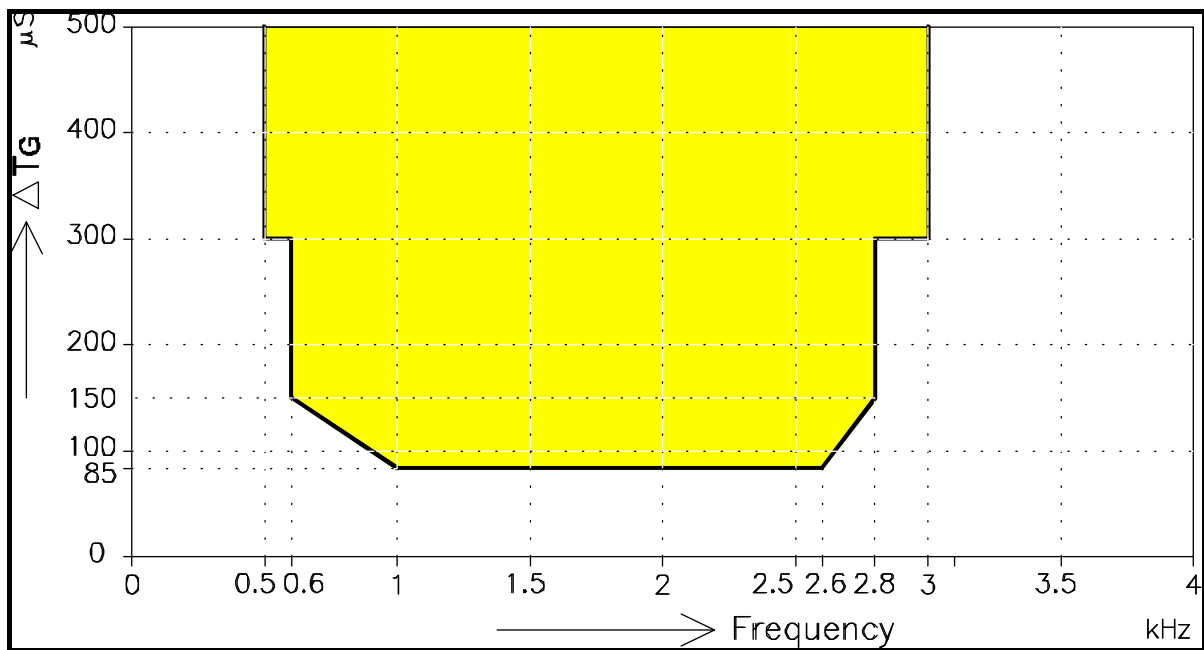


Figure 36: Group Delay Distortion Transmit

## 12.4.8 Out-of-Band Signals at TIP/RING Receive

With an 0dBm0 out-of-band sine wave signal with a frequency of ( $\ll 100\text{Hz}$  or  $3.4\text{kHz}$  to  $100\text{kHz}$ ) applied to the analog input, the level of any resulting frequency component at the digital output will stay at least X dB below a 0dBm0, 1kHz sine wave reference signal at the analog input.<sup>2)</sup>

<sup>1</sup> R is switched on: reference point is at TGmin

HPR is switched off: reference point is at 1.5 kHz

<sup>2</sup> Poles at  $12\text{ kHz} \pm 150\text{ Hz}$  and  $16\text{ kHz} \pm 150\text{ Hz}$  will be provided

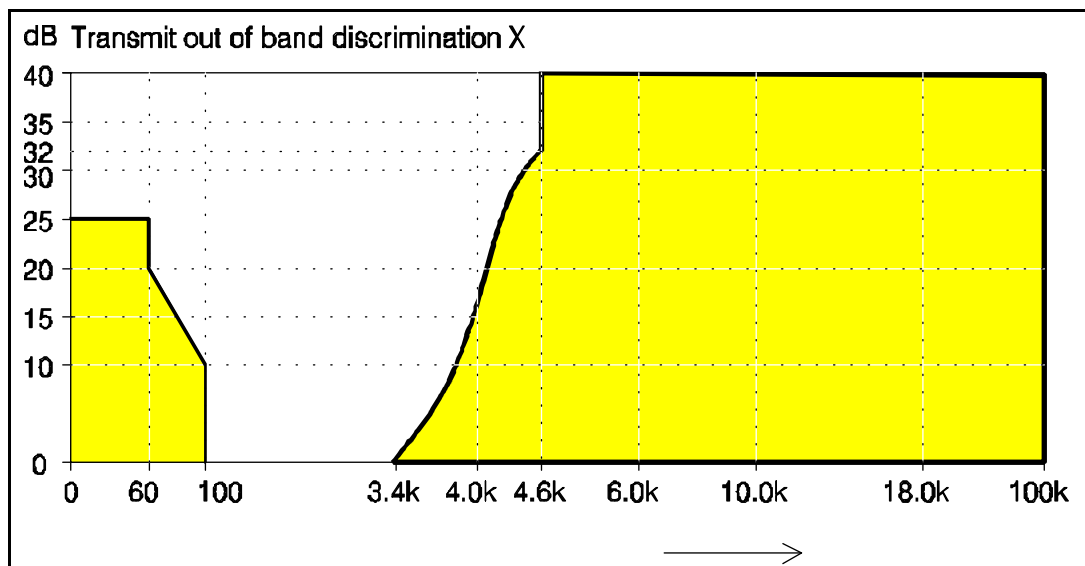


Figure 37: Out of Band Receive

#### 12.4.9 Out-of-Band Signals at TIP/RING Transmit

With a 0 dBm0 sine wave with a frequency of (300Hz to 3.99kHz) applied to the digital input, the level of any resulting out-of-band signal at the analog output will stay at least X dB below a 0 dBm0 1 kHz sine wave reference signal at the analog output.

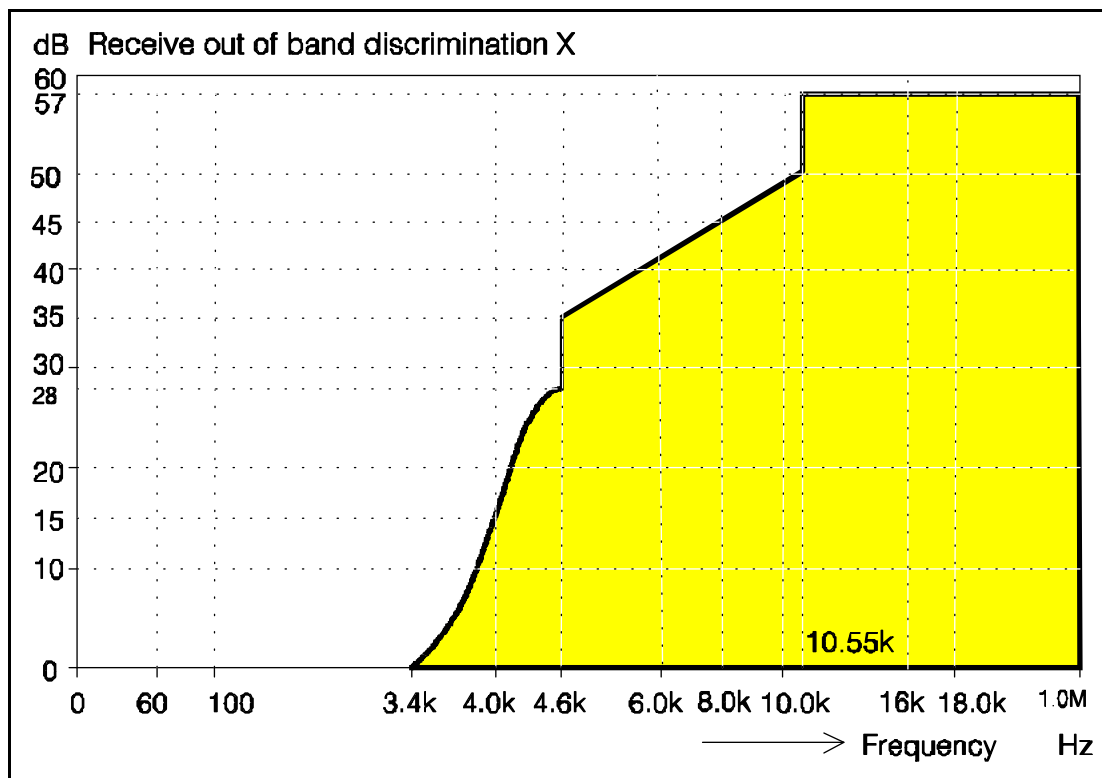


Figure 38: Out of Band Transmit

#### 12.4.10 Transhybrid Loss

The quality of Transhybrid-Balancing is very sensitive to deviations in gain and group delay - deviations inherent to ALIS A/D- and D/A-converters as well as to all external components used.

Measurement of ALIS Transhybrid-Loss: A 0dBm0 sine wave signal and a frequency in the range between 300 - 3400 Hz is applied to the digital input. The resulting analog output signal VOUT at TIP RING is received and cancelled by the TH Filter. The programmable filters FRR, AR, FRX, AX and IM and the balancing filter TH is enabled with coefficients optimised.

The resulting echo measured at the digital output is at least X dB below the level of the digital input signal as shown in the table below. (Filter coefficients will be provided)

Parameter	Symbol	Limit Values		Unit	Test condition
Transhybrid Loss at		min	typ		

## Electrical Performance Characteristic

300 Hz	THL 300	27	40	dB	TA=25° C; VDDA=4.25V;
500 Hz	THL 500	33	45	dB	
2500 Hz	THL2500	29	40	dB	
3000 Hz	THL3000	27	35	dB	
3400 Hz	THL3400	27	35	dB	

Table 30: Transhybrid Loss

The listed values for THL correspond to a typical variation of the signal amplitude and delay in the analog blocks.

Amplitude =typ.  $\pm 0.8$  dB

Delay =typ.  $\pm 0.5$   $\mu$ s

## 12.5 AC Timing Characteristics

### 12.5.1 Input/ Output Waveform for AC Tests

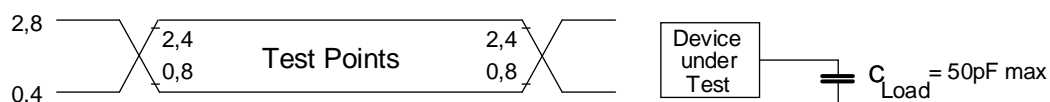


Figure 39: Waveform for AC Tests

### 12.5.2 Reset Timing

To reset ALIS to basic setting mode, negative pulses applied to the RESET pin have to be lower than 1.5 Volts (CMOS-Schmitt-Trigger Input) and longer than 500ns. Signals shorter than 100ns are ignored.

### 12.5.3 Control Interface Timing

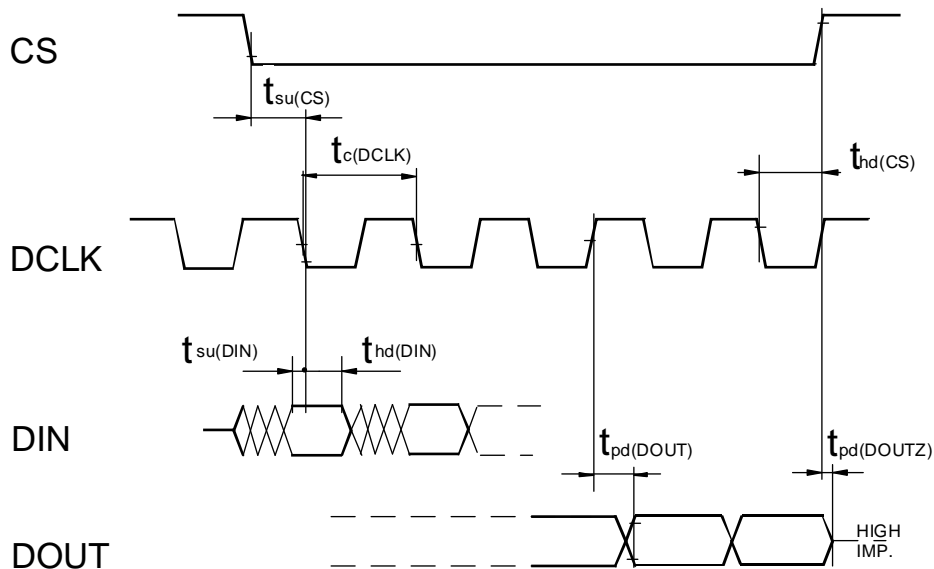


Figure 40: Control Interface Timing

VDD=VDDA= 5V± 5%; TA= 0 - 70°C

Parameter	Symbol	Limit Values			Unit
		min	typ	max	
Clock Cycle Time	$t_c(DCLK)$	1/1024		1	ms
Clock Duty Cycle		45	50	55	%
Setup Time, CS↓ before DCLK↓	$t_{su}(CS)$	50			ns
Hold Time, CS↑ after DCLK↓	$t_{hd}(CS)$	50			ns
Setup Time, DIN before DCLK↓	$t_{su}(DIN)$	50			ns
Hold Time, DIN after DCLK↓	$t_{hd}(DIN)$	50			ns
Delay Time, DCLK↑, to DOUT	$t_{pd}(DOUT)$			100	ns
Delay Time, CS↑ to DOUTZ	$t_{pd}(DOUTZ)$			100	ns

Table 31: Control Interface Switching Characteristics



### 12.5.4 Data Interface Timing

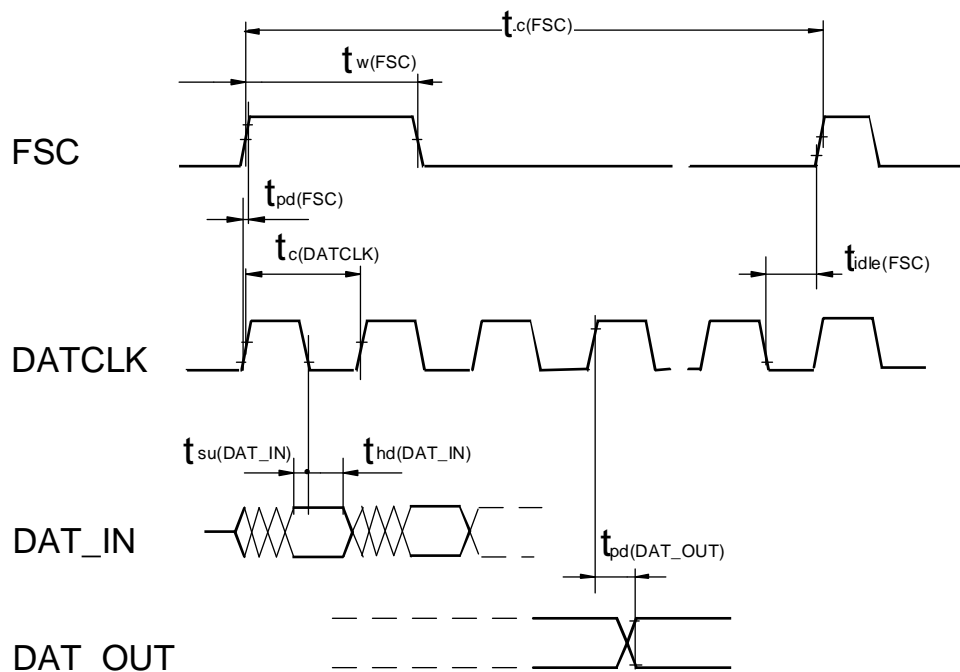


Figure 41: Data Interface Timing

VDD=VDDA= 5V $\pm$  5%; TA=0 - 70°C

Parameter	Symbol	Limit Values			Unit
		min	typ	max	
Data Clock Cycle Time	$t_c(\text{DATCLK})$	1/ 1024		1/ 128	ms
Data Clock Duty Cycle		45	50	55	%
Frame Synch Clock Cycle Time	$t_c(\text{FSC})$		125		$\mu\text{s}$
FSC Pulse Width (as Input)	$t_w(\text{FSC})$	500			ns
FSC Pulse Width (as Output)	$t_w(\text{FSC})$		1.4		$\mu\text{s}$
Setup Time, DAT_IN before DATCLK $\downarrow$	$t_{su}(\text{DAT\_IN})$	50			ns
Hold Time, DAT_IN after DATCLK $\downarrow$	$t_{hd}(\text{DAT\_IN})$	50			ns
Delay Time, DATCLK $\uparrow$ to DAT_OUT	$t_{pd}(\text{DAT\_OUT})$			100	ns
Idle Time, DATCLK $\downarrow$ to FSC $\uparrow$	$t_{idle}(\text{FSC})$	488			ns
Delay Time, DATCLK $\uparrow$ to FSC $\uparrow$	$t_{pd}(\text{FSC})$	0			ns

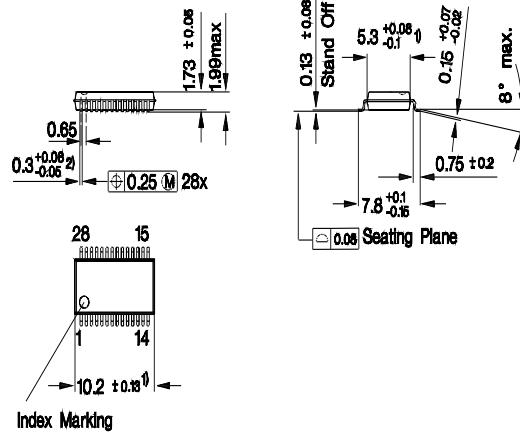
Table 32: Data Interface Switching Characteristics

## Electrical Performance Characteristic

### 12.5.5 CPackage Outlines

#### P-SSOP28

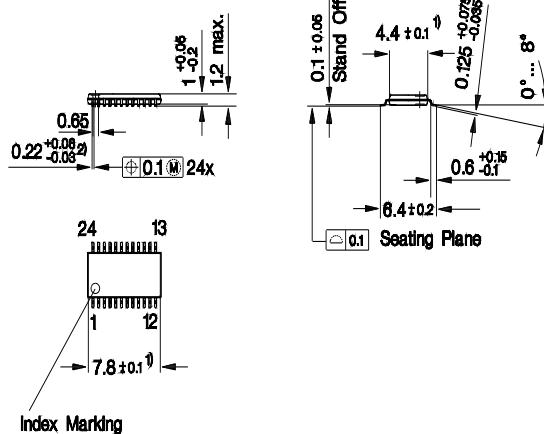
(Plastic Metric Quad Flat Package)



- 1) Does not include plastic or metal protrusion of 0.15 max. per side
- 2) Does not include dambar protrusion of 0.08 max. per side

#### P-TSSOP24

(Plastic Metric Quad Flat Package)



- 1) Does not include plastic or metal protrusion of 0.15 max. per side
- 2) Does not include dambar protrusion of 0.08 max. per side

Sorts of Packing  
Package outlines for tubes, trays etc. are contained in our  
Data Book "Package Information".

SMD = Surface Mounted Device

Dimensions in mm