ICs for Communications

Two Channel Codec Filter for Terminal Applications SICOFI**®**2-TE PSB 2132 Version 1.2 Four Channel Codec Filter for Terminal Applications SICOFI**®**4-TE PSB 2134 Version 1.2

Data Sheet 09.97

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PSB 2132 PSB 2134

Table of Contents Page

PSB 2132 PSB 2134

Table of Contents Page

1 Overview

The Signal Processing Codec Filter for terminal applications PSB 2132/4 SICOFI2/4-TE is a special derivative of the SIEMENS programmable codec-filter-IC family designed for terminal applications featuring two or four POTS interfaces.

It can be directly connected to the IOM-2 interface in terminal mode running at 1.536 MHz clock rate. PCM data is transfered using the bit clock signal at 768 kHz.

Programming of internal registers is done via the serial microcontroller interface.

Only two external capacitors per channel are needed to complete the functionality of the PSB 2132/4. The internal level accuracy is based on a very accurate bandgap reference. The frequency behaviour is mainly determined by digital filters, which do not have any fluctuations. As a result of the new ADC- and DAC- concepts linearity is only limited by second order parasitic effects. Although the device works only from one single 5 V supply there is a very good dynamic range available.

The PSB 2132/4 is a DSP based codec which allows the integration of filters and tone generators besides the regular A- or u-law conversion. In addition it integrates I/O extentions to the microcontroller and provides the necessary I/O pins to control the SLIC or discrete SLIC replacement. Interrupts are generated to the microcontroller if changes (e.g. Off-Hook detection) have been occured. The PSB 2132/4 provides a ring frequency output pin. This pin has a programmable clock frequency to meet the European and US ringing frequency requirements using only one external divider.

The IOM-2 data lines DU and DD can both be used for transmitting or receiving voice data. The position of each receive and transmit timeslot is programmable. Internal communication between analog ports is supported by programming each channel to the same timeslot but reversing the data lines. Thus the transmitted PCM data is transmitted by one port and received by the second port via the same timeslot. An additional IC for switch matrix is eliminated.

The PSB 2132/4 is specially of interest for applications, which need to serve different country specific characteristics on the POTS interface. Since all filters are programmable, adaptation to these country specific requirements may be done only by software parameters using the same hardware.

Preliminary Data CMOS

1.1 Features

- Single chip programmable CODEC and FILTER to handle two or four POTS interfaces
- IOM-2 compatible interface (1.536 MHz DCL, 768 kHz Bit clock)
- Internal communication between POTS interfaces
- Programmable I/O lines for signaling information per channel
- Programmable ring generator output
- Two programmable tone generators per channel
- Serial microcontroller interface
- Digital signal processing technique
- High analog driving capability (300 Ω) for direct driving of transformers
- Programmable digital filters to adapt the transmission behaviour especially for
	- AC impedance matching
	- transhybrid balancing
	- frequency response
	- gain
	- A/µ-law conversion
- Single 5 V power supply
- Low power 0.9 μ m analog CMOS technology
- Advanced test capabilities
- P-MQFP-64 package

PSB 2132 PSB 2134

Overview

1.2 Pin Configuration

(top view)

Figure 1

Overview

1.3 Pin Definition and Functions

Common Pins for all Channels

PSB 2132 PSB 2134

Overview

Specific Pins for Channel 1

Specific Pins for Channel 2

Specific Pins for Channel 3 (PSB 2134 only)

Specific Pins for Channel 4 (PSB 2134 only)

 $1)$ A 100 nF cap. should be used for blocking these pins, see also on page 83

²⁾ The value for the capacitor needed, depends on the input impedance of the 'SLIC'-circuitry. For choosing the appropriate values see figure on page 72.

2 Functional Description

2.1 System Integration

The SICOFI2/4-TE is connected to an IOM-2 compatible transceiver such as the PEB 8191 INTC-Q for U-interface or NT-applications or the PSB 2186 ISAC-S TE or PSB 2115 IPAC for S/T-interface applications. The FSC output is connected to the FSC input on the SICOFI2/4-TE. The DCL output of the transceiver is fed to the DCL input of the SICOFI2/4-TE which is used as master clock. For transferring PCM data, the bit clock signal of the transceiver (BCL, 768 kHz) is connected to the SICOFI2/4-TE.

Figure 2

Semiconductor Group 13 09.97

The microcontroller interface is connected to a microcontroller. Since the data transfer does not require duplex operation it can be connected both to SPI compatible microcontrollers (Siemens C5xx series, C161 series) as well as to Intel C51 based ones.

The SICOFI2/4-TE provides an high active interrupt output. The interrupts are caused by changes on the input lines of each channel. In order to operate it is necessary to keep DCL running all the time. If DCL is stopped in order to reduce the power consumption of the system, additional hardware is required. This hardware may be used to generate directly an interrupt to the microcontroller which may than request IOM-clocking.

Each channel serves seven I/O lines (2xO, 2xI, 3xI/O) which are used to control the inputs of the SLIC or to fed the outputs of a SLIC to the microcontroller.

The RGEN output can be used to generate the input signal of a ringing SLIC. Its frequency is programmable down to 35,7 Hz.

2.2 SICOFI®2/4-TE Principles

The SICOFI2/4-TE is designed for terminal adapters and Intelligent NT (NT1plus) applications.

It is designed to reduce the number of external components required for the integrated or discrete SLIC.

The SICOFI-2/4 TE bridges the gap between analog and digital voice signal transmission in modern telecommunication systems. High performance oversampling Analog-to-Digital Converters (ADC) and Digital-to-Analog Converters (DAC) provide the conversion accuracy required. Analog antialiasing prefilters (PREFI) and smoothing postfilters (POFI) are included. The connection between the ADC and the DAC (with high sampling rate) and the DSP, is done by specific Hardware Filters, for filtering like interpolation and decimation. The dedicated Digital Signal Processor (DSP) handles all the algorithms necessary e.g. for PCM bandpass filtering, sample rate conversion and PCM companding. The PCM-interface handles digital voice transmission, a serial µC-interface handles SICOFI2/4-TE feature control and transparent access to the SICOFI2/4-TE command and indication pins. To program the filters, precalculated sets of coefficients are downloaded from the system to the on-chip Coefficient-RAM (CRAM).

Figure 3 SICOFI®2/4-TE Signal Flow Graph (for any channel)

Transmit Path

The analog input signal has to be DC-free connected by an external capacitor because there is an internal virtual reference ground potential. After passing a simple antialiasing prefilter (PREFI) the voice signal is converted to a 1-bit digital data stream in the Sigma-Delta-converter. The first downsampling steps are done in fast running digital hardware filters. The following steps are implemented in the micro-code which has to be executed by the central Digital Signal Processor. This DSP-machine is able to handle the workload for all four channels. At the end the fully processed signal (flexibly programmed in many parameters) is transferred to the PCM- interface in a PCM-compressed signal representation.

Receive Path

The digital input signal is received via the PCM interface. Expansion, PCM-Law-pass-filtering, gain correction and frequency response correction are the next steps which are done by the DSP-machine. The upsampling interpolation steps are again processed by fast hardware structures to reduce the DSP-workload. The

upsampled 1-bit data stream is then converted to an analog equivalent which is smoothed by a POST-Filter (POFI). As the signal V_{OUT} is also referenced to an internal virtual ground potential, an external capacitor is required for DC-decoupling.

Loops

There are two loops implemented. The first is to generate the AC-input impedance (IM) and the second is to perform a proper hybrid balancing (TH). A simple extra path IM2 (from the transmit to the receive path) supports the impedance matching function.

Test Features

There are four analog and five digital test loops implemented in the SICOFI-2/4 TE. For special tests it is possible to cut off the receive and the transmit path at two different points. In addition, external test loops including the subscriber line measurement are possible using the level meatering feature.

2.3 The IOM-2 PCM-interface

One serial PCM-interface is used for transfer of A- or µ-law compressed voice data. The PCM-interface consists of 4 pins:

- BCL: IOM-2 bit clock, 768 kHz
- FSC: Frame Synchronization Clock, 8 kHz
- DU: Data transmit or receive in data upstream direction
- DD: Data receive or transmit in data downstream direction

The Frame Sync FSC pulse identifies the beginning of a receive and transmit frame for all of the two / four channels. The BCL clock is the signal to synchronize the data transfer on both lines DU and DD. Bytes in all channels are serialized to 8 bit width and MSB first. As a default setting, the rising edge indicates the start of the bit, while the falling edge is used to latch the contents of the received data.

The data rate of the interface is fixed to 768 kHz. A frame consists of 12 time slots of 8 bits each. In the Time Slot Configuration Registers CR5 and CR6 the user can select an individual time slot, and one of two data lines, for any of the voice channels. Receive and transmit time slots can also be programmed individually. An extra delay of up to 7 clocks, valid for all channels, as well as the sampling slope may be programmed (see XR6).

A typical example is shown below.

Functional Description

Figure 5 Example for IOM-2 Terminal Mode

2.4 The µ**-Controller Interface**

The internal configuration registers, the signaling interface, and the Coefficient-RAM (CRAM) of the SICOFI-2/4-TE are programmable via a serial μ -Controller interface.

The u-Controller interface consists of four lines: CS, DCLK, DIN and DOUT:

CS is used to start a serial access to the SICOFI-2/4-TE registers and 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.

Figure 6

Example for a Write Access, with Two Data Bytes Transferred

If the first eight bits received via DIN specify a read-command, the SICOFI-2/4 TE will start a response via DOUT with its specific address byte (81_H). After transmitting this identification, the specified n data bytes (contents of configuration registers, or contents of the CRAM) will follow on DOUT.

Functional Description

Figure 7 Example for a Read Access, with One Data Byte Transferred via DOUT

The data transfer is synchronized by the DCLK input. The contents of DIN is latched at the rising edge of DCLK, while DOUT changes with the falling edge of DCLK. During execution of commands that are followed by output data (read commands), the device will not accept any new command via DIN. The data transfer sequence is completed by setting \overline{CS} to high.

To reduce the number of connections to the μ P DIN and DOUT may be strapped together, and form a bi-directional data-'pin'.

For special applications a byte by byte transfer is needed. This can be done by prolonging the high time of DCLK for a user defined 'waiting time' after transferring any byte.

Figure 8

Example for a Write/Read Access, with a Byte by Byte Transfer, and DIN and DOUT Strapped Together

The Identification Byte is "81 $_H$ " for the PSB 2132/34.

2.5 The Signaling Interface

The SICOFI-2/4 TE signaling interface is made up of 2 input pins (SIx_0, SIx_1), two output pins (SOx_0, SOx_1) and three bi-directional programmable pins (SBx_0, SBx_1, SBx_2) per channel.

Figure 9

The purpose of these pins is to control the SLIC functions without additional ports on the host or microcontroller.

Functional Description

Figure 10

The status bits of all SIx_0 and SIx_1 inputs are stored in the XR0 register (RD). Similar the control bits of SOx_0 and SOx_1 are stored in the XR0 register (WR).

The bidirection status bits are arranged such that all SBx 1 and SBx 0 bits are controlled / read via the XR1 register. The correspondig direction register is the XR2 register. The third bidirectional status bit of each channel is accessed via the four most significant bits of the XR3 register while the least significant four bits specify the corresponding direction.

Figure 11

Depending on the application, the lines can be group individually to support the best software interface. E.g. if a DTMF receiver is connected to the SICOFI2/4-TE, the pins SB2_1, SB2_0,SB1_1,SB1_0 may be used for the data bus. This simplifies the software since the value can be read directly from the register.

Figure 12

Additional two interrupt pins (INT12, INT34) are provided. If one of the input pins for channel 1 or 2, or one of the bi-directional pins for channel 1 and 2 (if programmed as inputs) changes, and being stable for the debounce time specified in Register XR4, INT12 will go from '0' to '1'. This interrupt is cleared if the appropriate registers (XR0, $XR1$ and $XR3$) are read via the serial μ C-interface. Pin INT34 provides the same functionality for channel 3 and 4.

2.6 Ring Generator and Special Purpose Pin

For special purposes two additional output signals are provided by the SICOFI-2/4 TE.

RGEN (see also register XR4) will provide a programmable ring generator output of 2 to 28 ms. The output of RGEN diveded by four can be used to drive the ring input of a ringin SLIC. RGEN delivers a square-wave signal (duty cycle 1:1).

CHCLK will provide 3 different frequencies (256 kHz, 512 kHz or 16384 kHz). Both signals are only available if a valid signal is applied to the DCL-pin.

3 Programming the SICOFI® -2/4-TE

With the appropriate commands, the SICOFI2/4-TE can be programmed and verified very flexibly via the µ-Controller interface.

With the first byte received via DIN, one of 3 different types of commands (SOP, XOP and COP) is selected. Each of those can be used as a write or read command. Due to the extended SICOFI2/4-TE feature control facilities, SOP, COP and XOP commands contain additional information (e.g. number of subsequent bytes) for programming (write) and verifying (read) the SICOFI2/4-TE status.

A write command is followed by up to 8 bytes of data. The SICOFI2/4-TE responds to a read command with its specific identification and the requested information, that is up to 8 bytes of data.

3.1 Types of Command and Data Bytes

The 8-bit bytes have to be interpreted as either commands or status information stored in Configuration Registers or the Coefficient-RAM. There are three different types of SICOFI -2/4-TE commands which are selected by bit 3 and 4 as shown below.

SOP STATUS OPERATION: SICOFI2/4-TE status setting/monitoring

Note: 1) Command/Indication (signaling) channel.

Storage of Programming Information

3.2 Examples for SICOFI®2/4-TE Commands

SOP - Write Commands

XOP - Write Commands

COP - Write Commands

SOP - Read Commands

XOP-Read Commands

COP-Read Commands

Example of a Mixed Command

3.3 SOP Command

To modify or evaluate the SICOFI2/4-TE status, the contents of up to 6 configuration registers CR0 .. CR5 may be transferred to or from the SICOFI2/4-TE. This is started by a SOP-Command (status operation command).

All other codes are reserved for future use !

Note: If only one configuration register requires modification, for example CR5, this can be accomplished by setting LSEL = 101 and releasing pin CS after CR5 is written.

3.3.1 CR0 Configuration Register 0

Configuration register CR0 defines the basic SICOFI2/4-TE settings, which are: enabling/disabling the programmable digital filters.

3.3.2 CR1 Configuration Register 1

Configuration register CR1 selects tone generator modes and other operation modes.

¹⁾ Tone generator 2 is not available if Level Metering Function is enabled!

3.3.3 CR2 Configuration Register 2 $1)$ Explanation of the level metering function: Bit 7 and 2008 and 20 **COT/R 0 IDR LM LMR V+T COT/R** Selection of Cut off Transmit/Receive Paths 0 0 0: Normal Operation 0 0 1: COT16 Cut Off Transmit Path at 16 kHz (input of TH-Filter) 0 1 0: COT8 Cut Off Transmit Path at 8 kHz (input of compression, output is zero for µ-law, 1 LSB for A-law) 1 0 1: COR4M Cut Off Receive Path at 4 MHz (POFI-output) 1 1 0: COR64 Cut Off Receive Path at 64 kHz (IM-filter input) **IDR** Initialize Data RAM $IDR = 0$: Normal operation is selected IDR = 1: Contents of Data RAM is set to 0 (used for production test purposes) **LM** Level Metering function ¹⁾ $LM = 0$: Level metering function is disabled $LM = 1$: Level metering function is enabled **LMR** Result of Level Metering function (this bit can not be written) $LMR = 0$: Level detected was lower than the reference $LMR = 1$: Level detected was higher than the reference **V+T** Add Voice signal and Tone Generator signal $V+T = 0$: Voice or Tone Generator is fed to the DAC $V+T = 1$: Voice and Tone Generator Signals are added, and fed to the Digital to Analog Converter

A signal fed to A/µ-Law compression via AX- and HPX-filters (from a digital loop, or externally via VIN), is rectified, and the power is measured. If the power exceeds a certain value, loaded to XR7, bit LMR is set to '1'. The power of the incoming signal can be adjusted by AX-filters.

PSB 2132 PSB 2134

Programming the SICOFI® -2/4-TE

Figure 13 'CUT OFF's' and Loops

Semiconductor Group 36 09.97
3.3.4 CR3 Configuration Register 3

²⁾ In this case the receive-path signal is attenuated 0.12 dB

3.3.5 CR4 Configuration Register 4

Configuration register CR4, sets the receiving time slot and the receiving PCM-highway.

3.3.6 CR5 Configuration Register 5

Configuration register CR5, sets the transmit time slot and the transmit PCM-highway.

3.4 COP Command

With a COP command coefficients for the programmable filters can be written to the SICOFI-2/4-TE coefficient-RAM or read from the Coefficient-RAM via the µ-Controller interface for verification

How to Program the Filter Coefficients

TH-Filter: Two (Four) sets of TH-filter coefficients can be loaded to the SICOFI2 (/4)-TE. Each sets can be selected for any of the two / four SICOFI2/4-TE channels, by setting the value of TH-Sel in configuration register CR0. Coefficient set 1 is loaded to the SICOFI2/4-TE via channel 1, set 2 is loaded via channel 2 and so on. For the SICOFI2-TE, only set 1 and 2 are available.

AX, AR, IM/R1, FRX, FRR-Filter, Tone-Generators:

An individual coefficient set is available for each of the two / four channels.

3.5 XOP Command

With the XOP command the SICOFI2/4-TE digital command/indication interface to a SLIC is configured and evaluated. Also other common functions are assigned with this command.

RST Software Reset

(same as RESET-pin, valid for all 2/4 channels)

 $RST = 1$: Reset

 $RST = 0$: No operation

- **RW** Read / Write Information: Enables reading from the SICOFI-2/4-TE or writing information to the SICOFI2/4-TE
	- $RW = 0$: Write to SICOFI2/4-TE
	- RW = 1: Read from SICOFI2/4-TE

LSEL Length select information, for setting 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 $(XR7, XR6, XR5, XR4, XR3, YR4)$ XR2, XR1, XR0)

Note: All other codes are reserved for future use! If only one configuration register requires modification, for example XR5, this can be accomplished by setting LSEL =101 and releasing pin CS after XR5 is written.

3.5.1 XR0 Extended Register 0

The signaling connection between SICOFI2/4-TE and a SLIC is performed by master device the SICOFI2/4-TE signaling input and output pins and Configuration Register XR0... XR4. Data received from the upstream master device are transferred to signaling output pins (SO, SB). Data at the signaling input pins are transferred to the upstream controller.

In Connection with XOP-Read Commands

PSB 2134 only

In Connection with XOP-Write Commands

PSB 2134 only

Common

- **SO2_1** Pin SO2_1 is set to the assigned value
- **SO2_0** Pin SO2_0 is set to the assigned value
- **SO1_1** Pin SO1_1 is set to the assigned value
- **SO1_0** Pin SO1_0 is set to the assigned value

3.5.2 XR1 Extended Register 1

This register transfers information to or from the programmable signaling pins.

In Connection with a XOP-Read Command

In Connection with a XOP-Write Command

PSB 2134 only

Note: After a 'Reset' of the device, all programmable pins are input pins!

3.5.3 XR2 Extended Register 2

This register controls the direction of the programmable signaling pins.

PSB1_0 = 1: Pin SB1_0 is command output

Note: After a 'Reset' of the device, all programmable pins are input pins!

3.5.4 XR3 Extended Register 3

This register transfers information to or from the programmable signaling pins and configures these pins.

In Connection with a XOP-Read Command

PSB 2134 only

In Connection with a XOP-Write Command

PSB 2134 only

PSB3_2 Programmable bi-directional signaling pin SB3_2 is programmed

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Programming the SICOFI® -2/4-TE

Note: After a 'Reset' of the device, all programmable pins are input pins!

3.5.5 XR4 Extended Register 4

Register XR4 provides two optional functions: debouncing of signaling input changes, and the configuration of the programmable ring generator output pin RGEN.

Signaling Debounce Interval N

To restrict the rate of changes on signaling input pins transferred, deglitching of the status information from the SLIC may be applied. New status information will be read into registers XR0, XR1, XR2 and XR3, and an interrupt on pin INT12 (INT34) will be generated, after it has been stable for N milliseconds. N is programmable in the range of 2 to 26ms in steps of 2 ms, with $N = 0$ the debouncing is disabled. The last two bit combinations are reserved for future use.

Configuration of RGEN

3.5.6 XR5 Extended Register 5

This register contains additional configuration items valid for all 2/4 channels

is '01' for Version 1.2

 $1)$ A crash occurs, if 2 or more channels are programed to transmit (talk) in the same time slot on the same line. In this case the crash-bit will be set, and transmission will be disabled for all affected channels.

3.5.7 XR6 Extended Register 6

This register configures the operation of the PCM-interface

X-S Transmit Slope

3.5.8 XR7 Extended Register 7

This register contains the 8-bit offset value for the level metering function

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Programming the SICOFI® -2/4-TE

3.5.9 Setting of Slopes in Register XR6

3.6 Operating Modes

Figure 16

3.6.1 RESET (Basic Setting Mode)

Upon initial application of V_{DD} or resetting pin RESET to '0' during operation, or by software-reset (see XOP command), the SICOFI2/4-TE enters a basic setting mode. Basic setting means, that the SICOFI2/4-TE configuration registers CR0... CR6 and XR0... XR7 are initialized to '0' for all channels.

All programmable filters are disabled, all programmable command/indication pins are inputs. The two tone generators as well as any testmodes are disabled. There is no persistence checking. Receive signaling registers are cleared. DOUT-pin is in high impedance state, the analog outputs and the signaling outputs are forced to ground.

If any voltage is applied to any input-pin before initial application of V_{DD} , the SICOFI2/4-TE may not enter the basic setting mode. In this case it is necessary to reset the SICOFI2/4-TE or to initialize the SICOFI2/4-TE configuration registers to '0'.

The SICOFI2/4-TE leaves this mode automatically after the RESET-pin is released.

3.6.2 Standby Mode

After releasing the RESET-pin, (RESET-state), the SICOFI2/4-TE will enter the Standby mode. The SICOFI2/4-TE is forced to standby mode with the PU-bit set to '0' in the CR1-register (POWERDOWN). All 2/4 channels must be programmed separately. During standby mode the serial SICOFI2/4-TE μ -Controller interface is ready to receive and transmit commands and data. Received voice data on DU, DD-pin will be ignored. SICOFI2/4-TE configuration registers and Coefficient-RAM can be loaded and read back in this mode. Data on signaling input pins can be read via the μ -Controller interface.

3.6.3 Active Mode (Power Up)

The operating mode for any of the four channels is entered upon recognition of a PU-bit set to '1' in a CR1-register for the specific channel.

3.6.4 Programmable Filters

Based on an advanced digital filter concept, the SICOFI-2/4 TE provides excellent transmission performance and high flexibility. The new filter concept leads to a maximum independence between the different filter blocks.

Impedance Matching Filter

– Easy selection between four different downloaded coefficient sets

Filters for Frequency Response Correction

- For line equalization and compensation of attenuation distortion
- Improvement of Group-Delay-Distortion by using minimum phase filters (instead of linear phase filters)
- FRR filter for correction of receive path distortion – 5 TAP programmable FIR filter operating at 8 kHz (60 bit) • FRX filter for correction of transmit path distortion – 5 TAP programmable FIR filter operating at 8 kHz (60 bit)
- Frequency response better 0.1 dB

Amplification/Attenuation -Filters AX1, AX2, AR1, AR2

- Improved level adjustment for transmit and receive
- Two separate filters at each direction for
	- Improved trans-hybrid balancing
	- Optimal adjustment of digital dynamic range
	- Gain adjustments independent of TH-filter

Amplification/Attenuation Receive (AR1, AR2)-Filter

Amplification/Attenuation Transmit (AX1, AX2)-Filter

3.6.5 QSICOS Software

The QSICOS-software has been developed to help to obtain an optimized set of coefficients both quickly and easily. The QSICOS program runs on any PC with at least 575 Kbytes of memory. This also requires MS-DOS Version 5.0 or higher, as well as extended memory.

Figure 17

QSICOS Supports:

• Calculation of Coefficients for the SICOFI2/4-TE

- Impedance Filter (IM) for return loss calculation (please note that the IM filter coefficients are different for the SICOFI2/4-TE and for the PEB 2465. QSICOS calculates the programming bytes for the SICOFI-4 IOM version PEB 2465. These bytes have to be converted with an additional tool to get the required SICOFI-2/4 TE programming bytes. The conversion tool QSUCCONV.EXE is part of the QSICOS software package.)
- FRR and FRX-filters for frequency response in receive and transmit path
- AR1, AR2 and AX1, AX2-filter for level adjustment in receive and transmit path
- Transhybrid Balancing Filter (TH) and
- Two programmable tone generators (TG 1 and TG 2)
- **Simulation of the SICOFI-2/4 TE and SLIC System** with fixed filter coefficients allows simulations of tolerances which may be caused e.g. by discrete external components.
- **Graphical Output of Transfer Functions to the Screen** for
	- Return Loss
	- Frequency responses in receive and transmit path
	- Transhybrid Loss
- **Calculation of the SICOFI-2/4 TE and SLIC system Stability**. The IM-filter of the SICOFI-2/4 TE adjust the total system impedance by making a feedback loop. Because the line is also a part of the total system, a very robust method has to used to avoid oscillations and to ensure system stability. The input impedance of the

SICOFI-2/4 TE and SLIC combination is calculated. If the real part of the system input impedance is positive, the total system stability can be guaranteed.

In addition to the individual calculation of coefficient sets Siemens will provide ready to use coefficient sets for selected SLICs.

Please contact your Siemens office for available information.

4 Transmission Characteristics

The figures in this specification are based on the subscriber-line board requirements. The proper adjustment of the programmable filters (transhybrid balancing, impedance matching, frequency-response correction) requires a complete knowledge of the SICOFI-2/4 TE's analog environment. Unless otherwise stated, the transmission characteristics are guaranteed within the test conditions.

Test Conditions

 $T_{\rm A}$ = 0 °C to 70 °C; $V_{\rm DD}$ = 5 V \pm 5%; GNDA1..4 = GNDD = 0 V R ¹) > 300 Ω; C _L < 50 pF; H(IM) = H(TH) = 0; H(R1) = H(FRX) = H(FRR) = 1; HPR and HPX enabled; $AR^{2}= 0$ to $- 8$ dB $AX^{3}=0$ to 8 dB for A-Law, 0 to 6 dB for μ -Law *f* = 1014 Hz; 0 dBm0; A-Law or µ-Law; $AGX = 0$ dB, 6.02 dB, $AGR = 0$ dB, -6.02 dB;

A-Law

A 0 dBm0 signal is equivalent to 1.095 Vrms. A + 3.14 dBm0 signal is equivalent to 1.57 Vrms which corresponds to the overload point of 2.223 V.

When the gain in the receive path is set at 0 dB, an 1014 Hz PCM sinewave input with a level 0 dBm0 will correspond to a voltage of 1.095 Vrms at the analog output.

When the gain in the transmit path is set at 0 dB, an 1014 Hz sine wave signal with a voltage of 1.095Vrms A-Law will correspond to a level of 0 dBm0 at the PCM output.

µ**-Law**

In transmit direction for μ -law an additional gain of 1.94 dB is implemented automatically, in the companding block (CMP). This additional gain has to be considered at all gain calculations, and reduces possible AX -gain from 8 dB (with A-Law) to 6 dB (with μ -Law)

A 0 dBm0⁴⁾ signal is equivalent to 1.0906 Vrms. A $+$ 3.17 dBm0 signal is equivalent to 1.57 Vrms which corresponds to the overload point of 2.223 V.

When the gain in the receive path is set at 0 dB, an 1014 Hz PCM sinewave input with a level 0 dBm0 will correspond to a voltage of 1.0906 Vrms at the analog output.

When the gain in the transmit path is set at 0 dB, an 1014 Hz sine wave signal with a voltage of 1.0906 Vrms will correspond to a level of 1.94 dBm0 at the PCM output.

 $\frac{1}{n}$ R_{L} , C_{L} forms the load on VOUT

²⁾ Consider, in a complete system, $AR = AR1 + AR2 + FRR + R1$

³⁾ Consider, in a complete system, $AX = AX1 + AX2 + FRX$

 $4)$ The absolute power level in decibels referred to (a point of zero relative level) the PCM interface levels.

Transmission Characteristics

¹⁾ Using equal-level, 4-tone method (EIA) at a composite level of $-$ 13 dBm0 with frequencies in the range between 300 Hz and 3400 Hz.

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Transmission Characteristics

4.1 Frequency Response

Figure 18 Receive: Reference Frequency 1014 Hz, Input Signal Level 0 dBm0

4.2 Group Delay

Maximum delays when the SICOFI2/4-TE 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.

Figure 20 Group Delay Distortion Transmit: Input Signal Level 0 dBm0

Figure 21 Group Delay Distortion Receive: Input Signal Level 0 dBm0 1)

¹⁾ HPR is switched on: reference point is at t_{Gmin} HPR is switched off: reference is at 1.5 kHz

4.3 Out-of-Band Signals at Analog Input

With an 0 dBm0 out-of-band sine wave signal with frequency $f \ll 100$ Hz or 3.4 kHz to 100 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 0 dBm0, 1 kHz sine wave reference signal at the analog input.¹⁾

Figure 22

¹⁾ Poles at 12 kHz \pm 150 Hz and 16 kHz \pm 150 Hz are provided

4.4 Out-of-Band Signals at Analog Output

With a 0 dBm0 sine wave with frequency *f* (300 Hz to 3.99 kHz) 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 23

4.5 Out of Band Idle Channel Noise at Analog Output

With an idle code applied to the digital input, the level of any resulting out-of-band power spectral density (measured with 3 kHz bandwidth) at the analog output, will be not greater than the limit curve shown in the figure below.

Figure 24

4.6 Overload Compression

4.7 Gain Tracking (receive or transmit)

The gain deviations stay within the limits in the figures below.

Figure 26

Gain Tracking: (measured with sine wave $f = 1014$ Hz, reference level is 0 dBm0)

4.8 Total Distortion

The signal to distortion ratio exceeds the limits in the following figure (measured with sine wave).

Figure 27

Receive or Transmit: measured with sine wave *f* = 1014 Hz. (C-message weighted for µ-law, psophometricaly weighted for A-law)

4.9 Single Frequency Distortion

An input signal with its frequency swept between 0.3 to 3 kHz for the receive path, or 0 to 12 kHz for the transmit path, any generated output signal with other frequency than the input frequency will stay 28 dB below the maximum input level of 0 dBM0.

4.10 Transhybrid Loss

The quality of Transhybrid-Balancing is very sensitive to deviations in gain and group delay - deviations inherent to the SICOFI-2/4 TE A/D- and D/A-converters as well as to all external components used on a line card (SLIC, OP's etc.)

Measurement of SICOFI-2/4 TE Transhybrid-Loss: A 0 dBm0 sine wave signal and a frequency in the range between 300-3400 Hz is applied to the digital input. The resulting analog output signal at pin V_{OUT} is directly connected to V_{IN} , e.g. with the SICOFI-2/4 TE

testmode "Digital Loop Back via Analog Port". The programmable filters FRR, AR, FRX, AX and IM are disabled, the balancing filter TH is enabled with coefficients optimized for this configuration $(V_{\text{OUT}} = V_{\text{IN}})$.

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)

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

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Proposed Test Circuit

5 Proposed Test Circuit

Figure 28

Guidelines for Board-Design

6 Guidelines for Board-Design

6.1 Board Layout Recommendation

Keep in mind that inside the SICOFI-2/4 TE all the different V_{DD} -supplies are connected via the substrate of the chip, and the areas connected to different grounds are separated on chip.

- a) Separate all digital supply lines from analog supply lines as much as possible.
- b) Use a separate GND-connection for the capacitor which is filtering the reference voltage (220 nF ceramic-capacitor at V_{REF}).
- c) Don't use a common ground-plane under the SICOFI-2/4-TE.
- d) Use a large ground-plane (distant from the SICOFI-2/4-TE) and use three single ground lines for connecting the SICOFI-2/4-TE: one common analog ground, one digital ground, and a third for the 220 nF capacitor connected to V_{RFF} .

6.2 Filter Capacitors

- a) To achieve a good filtering for the high frequency band, place SMD ceramic-capacitors with 100 nF from V_{DDA12} , V_{DDA32} and V_{DDREF} to GNDA.
- b) One 100 nF SMD ceramic-capacitor is needed to filter the digital supply $(V_{\text{DDD}}$ to GNDD).
- c) Place all filter capacitors as close as possible to the SICOFI-2/4-TE (most important!!!).
- d) Use one central Tantalum-capacitor with about 1 μ F to 10 μ F to block V_{DD} to GND.

Programming the SICOFI2/4-TE Tone Generators

6.3 Example of a SICOFI-2/4-TE-board

Figure 29

7 Programming the SICOFI2/4-TE Tone Generators

Two independent tone generators are available per channel. Switching on/off the tone generators is done by a SOP-Command for CR1-register. The frequencies are programmed via a COP-Command, followed by the appropriate byte-sequence.

When one or both tone generators are switched on, the voice signal is switched off, if V+T=0 (CR2) for the selected voice channel. To make the generated signal sufficient for DTMF, a programmable bandpass-filter is included. The default frequency for both tone
Programming the SICOFI2/4-TE Tone Generators

generators is 1000 Hz. The QSICOS-program contains a program for generating coefficients for variable frequencies.

The following table shows sequences for programming both the tone generators and the bandpass-filters to select common used frequencies:

Table 1

 $1)$ OC is used for programming Tone Generator 1, in channel 1 0D is used for programming Tone Generator 2, in channel 1

The resulting signal amplitude can be set by transmitting the AR1 and AR2 filters. By switching a 'digital loop' the generated sine-wave signal can be fed to the transmit path.

8 Application Note: Level Metering

8.1 Introduction

The purpose of this application note is to describe the handling of the Level Metering Function and the facts that should be taken into account when using it.

The Level Metering Function is a feature which allows a self test of the chip and also a test of the connected circuitry i.e. SLIC, subscriber line and analog telephone. No external components are needed for this function.

8.2 Level Metering Block

Figure 33 shows the location of the Level Metering Function in the signal flow graph of one channel.

Figure 30 Block diagram of one SICOFI2/4-TE channel

The level metering function is always used together with a bandpass filter. The programming of this bandpass filter has to be done by programming the tone generator coefficients of the tone generator TG2. Due to of the bandpass filter only the power of a certain test signal is measured and disturbences originated from other signals are avoided.

After passing the bandpass filter of the Level Metering Function the test signal in the transmit path is rectified and the power of this signal is compared to the programmable offset value of the Level Metering Function. The reading of the CR2-bit LMR shows if the measured level of the test signal was higher or lower than the programmed offset value.

There is a single 8-bit offset register available for all 4 channels. This offset register can be accessed as XR7 with a XOP command. With the QSICOS utility program 'Calculate Level Metering Function' the programming byte for the register XR7 can be calculated. Another way is to use the table of appendix A.

By using the Level Metering Function in channel 2 (or 4) the channel 1 (or 3) has to be in operating mode.

8.3 Measuring a Level via the Level Metering Function

To find the value of an unknown level as fast as possible, the offset register should be programmed with the byte in the middle of the measuring range. The reading of the configuration register 2 bit LMR shows if the measured level is higher or lower than the programmed level.

If the measured level is higher than the reference, the offset register should be programmed with the byte in the middle between the highest valid byte and the previous used byte. Otherwise the byte in the middle between the lowest valid byte and the previous used byte should be programmed in the offset register.

Figure 2 shows the last three steps of such a procedure.

Figure 31

Procedure to find an unknown level with the Level Metering Function

By repeating the interpolation again and again two consecutive bytes will be found where bit LMR is high for the lower and low for the higher byte. The value of the unknown level is between the levels assigned to these two consecutive bytes.

Appendix B shows the programming file LMch1a.SUC with this procedure.

The procedure to find an unknown level is predesigned to be carried out by software.

The first valid LMR-bit is available 4 ms after enabling the level metering via setting bit LM. Then the LMR-bit is updated every 4 ms corresponding to the topical test signal and the stored offset byte. During the measurement time of 4ms the test signal has to be stable. That means for the above mentioned procedure to find an unknown level:

1. After programming the offset register and enabling the Level Metering Function, the software has to wait for at least 4 ms before accessing the first valid LMR-bit.

2. After programming the offset register again the software has to wait for at least 4 ms before accessing bit LMR.

3 After a change of the test level the software has to wait for at least 8 ms before accessing bit LMR.

8.4 Relative Measuring Precision

The bytes for the offset register are assigned to voltage values with a distance of about 0,02 Vrms. Since a level is measured by determining the bytes below and above, the distance between two consecutive bytes defines the relative measuring precision.

The maximum relative measuring error is about 0.02 Vrms.

The absolute measuring precision is depending on the gain tracking and described in the SICOFI2/4-TE data sheet.

8.5 Generating Tests Signals

In order to perform a measurement, an appropriate test signal is necessary. There are 3 different ways to create a test signal: built-in tone generators, test equipment PCM4 by Wandel and Goltermann or an external analog test source.

8.6 Tone Generators

Only tone generator 1 is available for level metering. By setting the LM bit in CR2 tone generator 2 is automatically switched off. But it is necessary to program both bandpass filters with the same coefficients because the coefficients of tone generator 2 determine also the bandpass filter for the Level Metering Function. Only with identically adjusted bandpass filters a precise level metering is possible.

A very simplified block diagram is shown in figure 35.

Figure 32 Test signal generated by tone generator 1

The attenuation and amplification filters of the SICOFI2/4-TE can be used to amplify or attenuate the level of the tone generator 1. With disabled filters the tone generator 1 sends a level of -4.5 dBm0.

In order to avoid test signal attenuation by the balancing filter TH, it has to be disabled. It is necesarry to switch off the voice in the respective channel during level metering with V+T bit of CR2.

No external components are required to use the Level Metering Function with the built-in tone generators. This can be very helpful in digital exchange systems. It is a task of the exchange software to switch from time to time the tone generators on and to measure the level via the Level Metering Function. So the actual state of all SICOFI2/4-TE channels and connected circuitry can be supervised very comfortably.

8.7 PCM4

For development of a new application the PCM4 by Wandel and Goltermann can also be very helpful to send and receive test signals. Please make sure that the configuration of the PCM4 corresponds to the SICOFI2/4-TE configuration, for instance that the same companding law is used.

Figure 36 shows a test configuration with a PCM4 device.

Figure 33 Test signal for level metering provided by PCM4

For a digital exchange system it is possible to use a PCM4 as a test signal generator as well as a test signal receiver. To do this, the PCM4 has to be connected to a PCM highway and the exchange has to switch the test information from the PCM4 to the respective SICOFI2/4-TE and back.

By using a PCM4 the expenditure for level metering measurement, level evaluating and preparation for software processing is higher and already done in the level metering block of the SICOFI2/4-TE. That is why the Level Metering Function is more helpful in digital exchange systems than a PCM4.

8.8 Analog Test Source

For test purposes an analog test signal can be applied to the SICOFI2/4-TE analog input. Figure 37 shows such a configuration.

To calculate the applied analog level, the gain settings of the SICOFI2/4-TE filters together with the 0 dBm0 reference voltage have to be taken into consideration.

Figure 34 Measurement of an unknown level

8.9 Loops

If the test signal is fed via tone generator 1 or via PCM4 in the receive path of the SICOFI2/4-TE a loop is necessary to feed the test signal back to the transmit path.

Digital loops are implemented in the SICOFI2/4-TE and can be activated by writing register CR4. After switching an internal loop the measured level is a representation of the internal attenuation and amplification via the filter blocks. Such self tests show whether the SICOFI2/4-TE is working or not and how it is adjusted.

In a line card application a SLIC is connected to each channel of the SICOFI2/4-TE. The SLIC together with a connected analog telephone creates a loop from the receive-path to the transmit path. SLIC, subscriber line and telephone have a special impedance according to their specification. If only one of them changes the value, the returned test signal will be changed und will indicate a change in the corresponding system condition. Therefore, changes of the telephone state, the subscriber line length or the SLIC can be supervised.

8.10 Application Examples

8.10.1 Supervision the State of a Subscriber Line

The configuration of an application example is shown in figure 38.

Figure 35 Test configuration

The TAS 2100 emulates subscriber line lengths from 0 to 6 kft in 1 kft increments. 1 kft is equivalent to 0.3 km.

The SICOFI2/4-TE is programmed with the file TEST.SUC for operation with the Harris-SLIC HC 5502 and the specification for Germany. The file TEST.SUC is a component part of the QSICOS software.

After that the SICOFI2/4-TE is programmed with the file LMch1b.SUC of appendix C. The task of this file is to program the bandpass filters at 1516 Hz and to increase the amplification by programming the AR and AX filters. With the second part of the file LMch1b.SUC the Level Metering Function is activated and the offset register is loaded. The read command reads out CR2 with bit LMR. Bit LMR shows if the level detected is higher or lower than the reference stored in the offset register.

By changing the line length with the Loop Emulator the levels of appendix D are measured by using the procedure described in paragraph 2.1. The values of the metered levels are depending on the line lengths.

That means, that it is possible to determine the state of a connected telephone as well as the subscriber line length very comfortably due to the help of the Level Metering Function and the built-in tone generators.

8.10.2 Improvement of Transhybrid Balancing

The Level Metering Function can also be used for an improvement of transhybrid balancing. This can be very useful after calculating the filter coefficients via QSICOS software and getting a too low transhybrid loss because of too long (or too short) subscriber lines. With the knowledge of the subscriber line impedance and the telephone impedance an optimization of the transhybrid loss is possible.

For it a configuration like in figure 38 is used. The transhybrid filter is enabled with bit TH=1 of configuration register 0.

The tone generator 1 is programmed for sending a test frequency in the range between 300 and 3400 Hz, e.g. 300 Hz. The via Level Metering Function measured level (dBm0) minus the send level (dBm0), minus the value RLR (dB) and plus the value RLX (dB) is equivalent to the negative transhybrid loss.

Transhybrid loss/dB = send level/dBm0 + RLR/dB + RLX/dB - measured level/dBm0

In order to get information about the transhybrid loss over the whole frequency band measurements at some other frequencies are necessary. In appendix E is a list of some frequencies and assigned tone generator coefficients.

If the transhybrid loss measurement results are to low, another coefficient set has to be used for the respective channel. Either a transhybrid filter set of another channel can be used or a new coefficient set can be written to the coefficient RAM.

For calculating a transhybrid filter coefficient set via QSICOS producing a high transhybrid loss the value of the subscriber line impedance und the telephone

impedance has to be known. Figure 39 shows a simplified model of a subscriber line with an analog telephone set.

Figure 36 A simplified model of a subscriber line and an analog telephone set

The subscriber line resistance and the subscriber line capacitance are depending on the cable type. Appendix F shows cable parameters of some cable types at 1 kHz. The cable parameters multiplied with the line length form the line resistance (RL) and the line capacitance (CL).

The input impedance of the telephone can be measured with an impedance analyzer. With parallel measuring mode the telephone resistance (RT) and the telephone capacitance (CT) in the off-hook state can be determined.

The sum of the telephone capacitance and the line capacitance forms the capacitance ZLC and is an input for QSICOS. The other inputs for QSICOS are RL (ZLR1) and RT (ZLR2). With these values QSICOS can calculate coefficients for a high transhybrid loss. Often the SICOFI2/4-TE is used with the same type of SLIC on all four channels. Therefore all four coefficient sets can be the same. But for different subscriber line lengths different transhybrid filter coefficients are necessary. For getting a high transhybrid loss it is useful to calculate TH-filter coefficient sets for 4 different subscriber line lengths and to store them in the coefficient RAM. With the help of the Level Metering Function the best coefficient set with the highest transhybrid loss can be selected for each channel.

So an improving of the transhybrid balancing by measuring the transhybrid loss and loading or selecting another coefficient set is possible. No external measuring devices are necessary.

8.11 Appendix

Appendix A: Assignment of measured level and byte for offset register XR7

Appendix B: File LMch1a.SUC

;SICOFI2/4-TE LEVEL METERING in channel 1, Version 1.0 by R.Kitze, January 1997 ;Configuration: SICOFI4-µC Board V1.1 STUT 2466, EVC50X Board, Harris SLIC-Board STUS ;5502 V2.0. ;Please run the file TEST.SUC (QSICOS package) first to configure the SICOFI2/4-TE. ;--- ;Programming of the tone generators 1 and 2, bandpass is set to 1516 Hz W 0 0C A5 53 61 56 W 0 0D A5 53 61 56 ;--- W 0 13 00 04 71 7C ; power up channel 1, TH is disabled and level metering is enabled ;The test level is -10.2 dBm0. ;--- ;STEP 1: W 0 1F 28 E0 80 OF FF FF FF FF ; LM offset byte = 28 (middle of the measuring range) R 0 32 (5) ; Read CR2, CR1, CR0 ;CR2=04 indicates that the measured level is lower than the reference ;--- ;STEP 2: W 0 1F 14 E0 80 OF FF FF FF FF ; LM offset byte $= 14$ (middle of the target range) R 0 32 (5) ; Read CR2, CR1, CR0 ;CR2=04 indicates that the measured level is lower than the reference ;--- ;STEP 3: W 0 1F 0A E0 80 0F FF FF FF FF; LM offset byte $=$ 0A (middle of the target range) R 0 32 (5) ; Read CR2, CR1, CR0 ;CR2=06 indicates that the measured level is higher than the reference ;--- ;STEP n: W 0 1F 0F E0 80 0F FF FF FF FF; LM offset byte $=$ 0F (middle of the target range) R 0 32 (5) ; Read CR2, CR1, CR0 ;CR2=06 indicates that the measured level is higher than the reference ;--- ;STEP n+1: W 0 1F 12 E0 80 OF FF FF FF FF; LM offset byte $= 12$ (middle of the target range) R 0 32 (5) ; Read CR2, CR1, CR0 ;CR2=04 indicates that the measured level is lower than the reference ;--- ;STEP n+2: W 0 1F 11 E0 80 OF FF FF FF FF; LM offset byte $= 11$ R 0 32 (5) ; Read CR2, CR1, CR0 ;CR2=06 indicates that the measured level is higher than the reference

;---

;Result: The measured level is between the levels -9.95 dBm0 (byte 12) and -10.45 dBm0 (byte 11).

Appendix C: File LMch1b.SUC

;Supervision the state of a subscriber line via LEVEL METERING in channel 1 ;Version 1.0 by R.Kitze, January 1997 ;Configuration: SICOFI4-µC Board V1.1 STUT 2466, EVC50X Board, Harris SLIC-Board STUS ;5502 V2.0. ;Please run the file TEST.SUC (QSICOS package) first to configure the SICOFI4-µC. ;--- ;programming of the tone generators 1 and 2, bandpass is set to 1516 Hz W 0 0C A5 53 61 56 W 0 0D A5 53 61 56 ;programming the amplification/attenuation filters to 2.3 dBm0 at PCM output (test level) W 0 08 4B 53 2A 56 W 0 09 DD B7 02 3A ;--- W 0 13 00 04 71 7C ; power up channel 1, TH is disabled and level metering is enabled ;--- ;STEP 1: The Loop Emulator is set to 0 kft. The telephone is in the off-hook state. ;--- ;STEP 1.1: W 0 1F 28 E0 80 OF FF FF FF FF ; LM offset byte $=$ 28 (middle of the measuring range) R 0 32 (5) ; Read CR2, CR1, CR0 ;CR2=06 indicates that the measured level is higher than the reference ;--- ;STEP 1.2: W 0 1F 3C E0 80 0F FF FF FF FF ; LM offset byte $=$ 3C (middle of the target range) R 0 32 (5) ; Read CR2, CR1, CR0 ;CR2=06 indicates that the measured level is higher than the reference ;--- ;STEP 1.3: W 0 1F 46 E0 80 0F FF FF FF FF ; LM offset byte $=$ 46 (middle of the target range) R 0 32 (5) ; Read CR2, CR1, CR0 ;CR2=06 indicates that the measured level is higher than the reference ;--- ;STEP 1.4: W 0 1F 4B E0 80 0F FF FF FF FF; LM offset byte = 4B (middle of the target range) R 0 32 (5) ; Read CR2, CR1, CR0 ;CR2=06 indicates that the measured level is higher than the reference ;--- ;STEP 1.5: W 0 1F 4E E0 80 OF FF FF FF FF; LM offset byte $=$ 4E (middle of the target range) R 0 32 (5) ; Read CR2, CR1, CR0 ;CR2=06 indicates that the measured level is higher than the reference

. .

Application Note: Level Metering

Appendix D: Measured levels for different line lengths

Subscriber Line: Cable configuration 0.4 mm = 26 AWG

The measured level is between the lower/higher level.

Appendix E: Tone generator coefficients (bandpass Q-factor = 2)

Appendix F: Cable parameters at 1 kHz

9 Electrical Characteristics

Absolute Maximum Ratings

Note: Stresses above those listed here may cause permanent damage to the device. Exposure to absolute maximum rating conditions for extended periods may affect device reliability.

Operating Range

 T_A = 0 to 70 °C; V_{DD} = 5 V \pm 5%; GNDD = 0 V; GNDA = 0 V

Note: In the operating range the functions given in the circuit description are fulfilled.

Digital Interface

 $T_A = 0$ to 70 °C; $V_{DD} = 5$ V ± 5%; GNDD = 0 V; GNDA = 0 V

Analog Interface

 $T_{\rm A}$ = 0 to 70 °C; $V_{\rm DD}$ = 5 V \pm 5%; GNDD = 0 V; GNDA = 0 V

9.1 Coupling Capacitors at the Analog Interface

In Transmit direction, a 39 nF capacitor has to be connected to V_{IN} -pins. To fulfil the frequency response requirement in Receive direction, the value of the coupling capacitor (C_{ext1}) needed, depends on the input resistance of the SLIC-circuitry (equals the Analog-Output-Load: R_{load}).

Figure 37

9.2 Reset Timing

To reset the SICOFI-2/4 TE to basic setting mode, negative pulses applied to pin RESET have to be lower than 1.2 V (TTL-Schmitt-Trigger Input) and have to be longer than 3 μ s. Spikes shorter than 1 us will be ignored.

9.3 PCM-Interface Timing

Figure 38 Single Clocking Mode

 $1)$ All delay times are made up by two components: an intrinsic time (min-time), caused by internal processings, and a second component caused by external circuitry (C-load)

9.4 µ**-Controller Interface Timing**

Figure 39

¹⁾ All delay times are made up by two components: an intrinsic time (min-time), caused by internal processings, and a second component caused by external circuitry (C-load)

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Electrical Characteristics

9.5 Signaling Interface

9.5.1 From the µ**C-interface to the SO/SB-pins (data downstream)**

 $1)$ All delay times are made up by two components: an intrinsic time (min-time), caused by internal processings, and a second component caused by external circuitry (C-load)

9.5.2 From the SI/SB-pins to the µ**C-interface (data upstream)**

There is no way specifying the time when data applied to SI-pins (and SB-pins if programmed as signaling input pins) is sampled by the SICOFI2/4-TE.

The time only depends on internal signals (16 MHz masterclock, and status of various counters), and there is no link to a low frequency external signal.

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Package Outlines

10 Package Outlines

Sorts of Packing Package outlines for tubes, trays etc. are contained in our Data Book "Package Information". SMD = Surface Mounted Device Dimensions in mm

Semiconductor Group 85 09.97