Digital Signal Processing

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Textbook

- Sanjit K. Mitra, **Digital Signal Processing: A Computer-Based Approach**, 4th edition

Classical DSP Books

- Alan V. Oppenheim and Ronald W. Schafer, **Discrete-time Signal Processing**, Pearson
- Lawrence R. Rabiner and Bernard Gold, **Theory and Application of Digital Signal Processing**, Prince Hall
- J. Proakis, D. Manolakis, **Digital Signal Processing** 4th ed., Prentice-Hall, 2006

Contents

- Discrete-time signals and systems in the time domain
- Discrete-time signals and systems in the transform domain
  - Discrete-time Fourier transform
  - Discrete Fourier transform
  - z-Transform
  - Frequency response
- DSP algorithms implementation – Fast Fourier Transform (FFT)
- Digital filter structures
- IIR and FIR filter design

Assessment

- Assignments: 5%
- Two Quiz: 15%
- Laboratories: 30%
- Final Exam: 50%

Journals & Conferences in DSP

- Journals:
  - IEEE Transactions on Signal Processing (TSP)
  - IEEE Transactions on Circuits and Systems I (TCASI)
  - IEEE Transactions on Circuits and System II (TCASII)
  - IEEE Signal Processing Letter (SPL)
  - Signal Processing (Elsevier)
  - EURASIP Journal on Applied Signal Processing
  - Digital Signal Processing
  - Circuits Systems and Signal Processing (CSSP)

Journals & Conferences in DSP

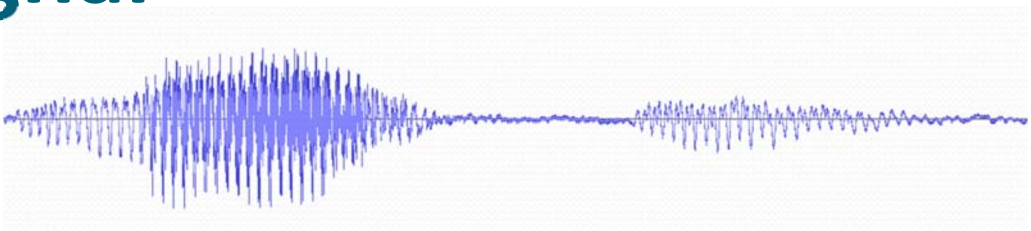
- Conference:
  - IEEE International Conference on Acoustic, Speech & Signal Processing (ICASSP)
  - IEEE International Symposium on Circuits & Systems (ISCAS)
  - European Signal Processing Conference (EUSIPCO)
  - International Conference on Digital Signal Processing (DSP)

Course Learning Outcomes

- CLO 1: I have an ability to **represent** discrete time signals and systems in time and frequency domain;
- CLO 2: I have an ability to **understand, represent, and analyse** linear time invariant discrete time systems in transformed domain by applying mathematics principles, such as **differential calculus, complex variables**.
- CLO 3: I have an ability to **analyse** digital filters and **design** digital filters to meet given specifications.
- CLO 4: I have an ability to **use a programming language** to conduct **analysis and design** of discrete-time signal processing systems to process discrete-time signals.

Lecture 1 Introduction

Signal



$f(t)$

$f(x, y)$

$f(t, x, y)$

Definition

- A signal can be defined simply as a mathematical function

$$y = f(x)$$

  where x is the **independent variable** which specifies the domain of the signal.

- $y=\sin(\omega t)$ is a function of a variable in the **time domain** and is thus a time signal;
- An image $I(x,y)$ is in the spatial domain.

Signal Processing

- A signal carries information.
- The objective of signal processing:
  - Interpretation and information extraction. (e.g. speech recognition, machine learning, etc.)
  - Convert one signal to another. (e.g. filter, generate control command, etc.)

- Signal Processing concerns with:
  - The mathematical representation of the signal
  - The algorithmic operation carried out on the signal

Representation of Signal

- In terms of basis functions in the domain of original independent variable,
  - Time
  - Spatial, etc., or

- In terms of basis functions in a transformed domain,
  - Discrete Fourier Transform
  - z transform, etc.

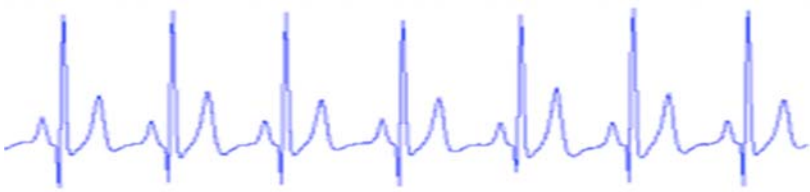
Classification of Signals

- Continuous vs. Discrete
- Real-valued vs. Complex-valued
- 1-D signal vs. 2-D signal vs. M-D signal
- 稳定 Stationary vs. Non-stationary
- etc.

均值与方差不变

Characterization of Signals

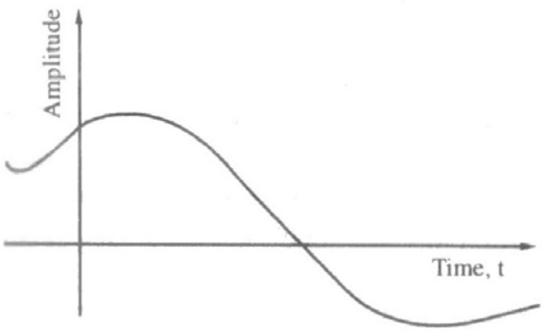
- The value of a signal at a specific value of the independent variable is called its **amplitude**. → 可正可负
- The variation of the amplitude as a function of the independent variable is called its **waveform**.
- Let’s consider 1-D signal
  - The independent variable is usually labeled as **time**.



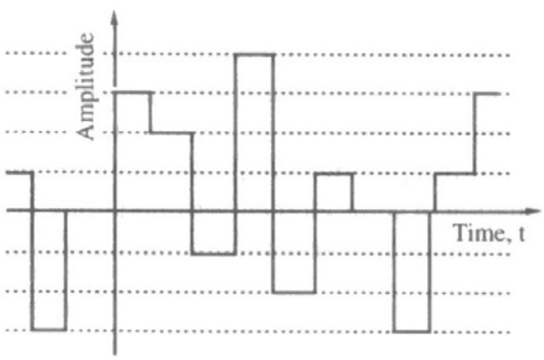
Continuous and Discrete Signals

- If the independent variable is continuous, the signal is called a **continuous-time (CT) signal**.
  - A continuous time signal is defined at every instant of time.
- If the independent variable is discrete, the signal is called a **discrete-time signal**.
  - A discrete time signal takes certain numerical values at specified discrete instants of time, and between these specified instants of time, the signal is **not defined**.
  - A discrete time signal is basically a sequence of numbers.

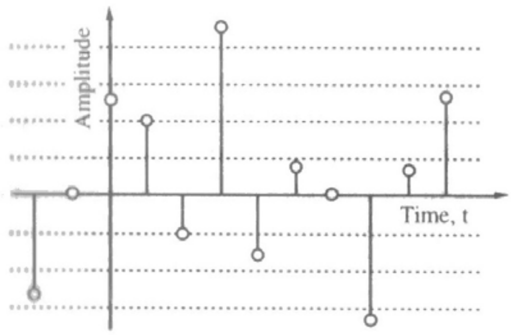
- A continuous-time signal with a continuous amplitude is usually called an **analog signal**.
  - A speech signal is an example of an analog signal.



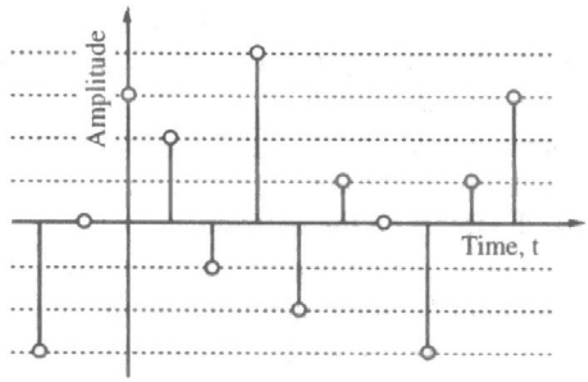
- A continuous-time signal with discrete valued amplitudes has been referred to as a **quantized boxcar signal**.
  - This type of signal occurs in digital electronic circuits where the signal is kept at fixed level (usually one of two values) between two instants of clocking.



- A discrete time signal with continuous valued amplitudes is called a **sampled-data signal**.
  - The amplitude of the signal may be any value.



- A discrete time signal with discrete valued amplitudes represented by a finite number of digits is referred to as a **digital signal**.
  - A digital signal is thus a quantized sampled-data signal.



Typical Signal Processing Operations

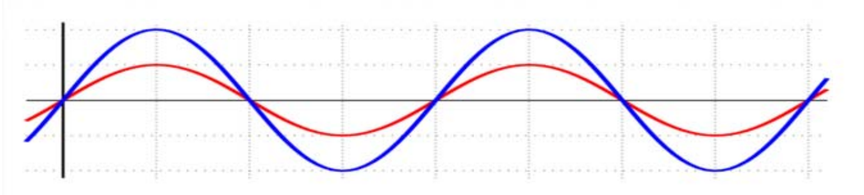
- In the case of analog signals, most signal processing operations are usually carried out in the **time domain**.
- In the case of discrete time signals, both **time domain and frequency domain** applications are employed.
- In either case, the desired operations are implemented by a combination of some **elementary operations** such as:
  - Simple time domain operations
  - Filtering
  - Amplitude modulation

Elementary Time-Domain Operations

- Three most basic time-domain signal operations
  - Scaling
  - Delay
  - Addition

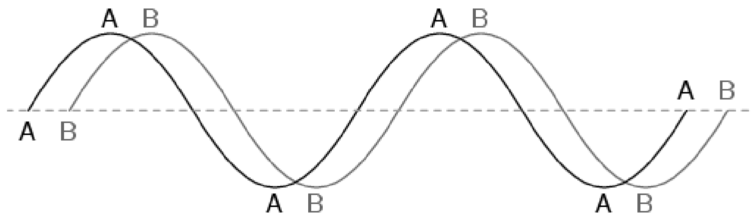
Scaling

- **Scaling** is simply the multiplication of a signal by a positive or a negative constant.
  - In the case of analog signal $x(t)$, the scaling operation generates a new signal $y(t) = \alpha x(t)$, where α is the multiplying constant.
  - The operation is called **amplification**, if $|\alpha| > 1$;
  - The operation is called **attenuation**, if $|\alpha| < 1$.



Delay

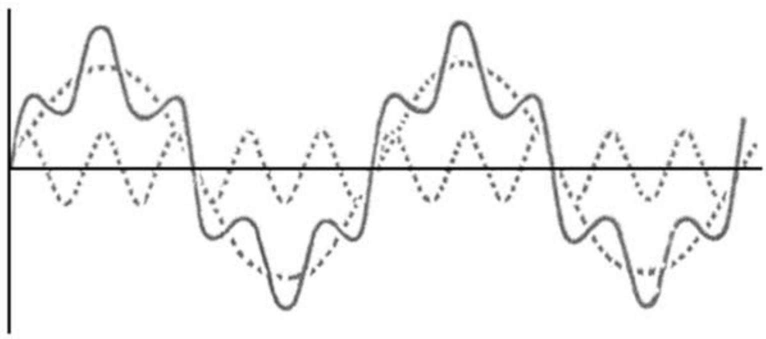
- **Delay** operation generates a signal that is delayed replica of the original signal.
  - In the case of analog signal $x(t)$, $y(t) = x(t - t_0)$ is the signal obtained by delaying $x(t)$ by the amount t_0, assuming $t_0 > 0$.



  - If $t_0 < 0$, it is an **advance** operation.

Addition

- **Addition** operation generates a new signal by the addition of signals. For instance, $y(t)=x_1(t)+x_2(t)$ is the signal generated by the addition of the three analog signals $x_1(t)$ and $x_2(t)$.



Why Learn DSP?

- Swiss-Army-Knife of modern EE
- Impacts all aspects of modern life
  - Communications (wireless, internet, GPS...)
  - Control and monitoring (cars, machines...)
  - Multimedia (mp3, cameras, videos, restoration ...)
  - Health (medical devices, imaging....)
  - Economy (stock market, prediction)
  - More....

DSP

DPS在信息技术中的地位和作用

信息获取 → 信息处理 → 信息传输 → 信息显示 → 信息应用

信息获取	信息处理	信息传输	信息显示	信息应用
光电测试技术 检测技术 传感器技术	DSP技术与应用 误差理论与数据处理 测试信号分析与处理	网络与通信技术 仪器总线技术 接口电路	数据可视化 虚拟仪器技术	导航与控制技术 机器人技术 传感网与物联网 智能仪器 控制技术与系统

DPS在信息技术中的地位和作用

 **中国制造2025（高质量发展）离不开DSP技术的支撑**

《中国制造2025》（高质量发展）是中国政府2015年3月提出的实施制造强国战略第一个十年的行动纲领。
围绕实现制造强国的战略目标，《中国制造2025》明确了9项战略任务和重点，提出了8个方面的战略支撑和保障。

五大工程：制造业创新中心建设工程、强化基础工程、智能制造工程、绿色制造工程、高端装备创新工程。
十个重点领域：新一代信息技术产业、高档数控机床和机器人、航空航天装备、海洋工程装备及高技术船舶、先进轨道交通装备、节能与新能源汽车、电力装备、农机装备、新材料、生物医药及高性能医疗器械。

Advantages of DSP

- Flexibility
- System/implementation does not age
- “Easy” implementation
- Reusable hardware
- Sophisticated processing
- Process on a computer
- (Today) Computation is cheaper and better

Example I: Audio Compression

感知的

- Compress audio by 10x without perceptual loss of quality.
- Sophisticated processing based on models of human perception
- 3MB files instead of 30MB -
  - Entire industry changed in less than 10 years!

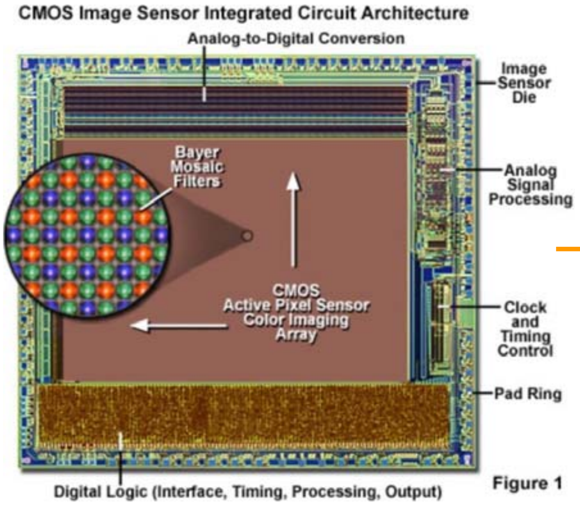
Historical Forms of Compression

- Morse code: dots (1 unit) Dashes (3 units)
  - Code Length inversely proportional to frequency
  - E (12.7%) = . (1 unit) Q (0.1%) = --.- (10 units)

- "92 Code" - Used by Western-Union in 1859 to reduce BW on telegraph lines by numerical codes for frequently used phrases
  - 1 = wait a minute
  - 73 = Best Regards
  - 88 = Love and Kisses

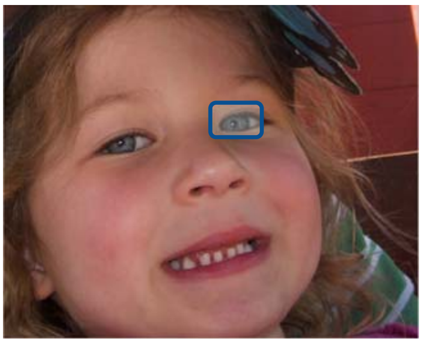
73	Best	Regards
--… …--	-… . … - /	.-. . --. .- .-. -.. …
19 units	59 units	

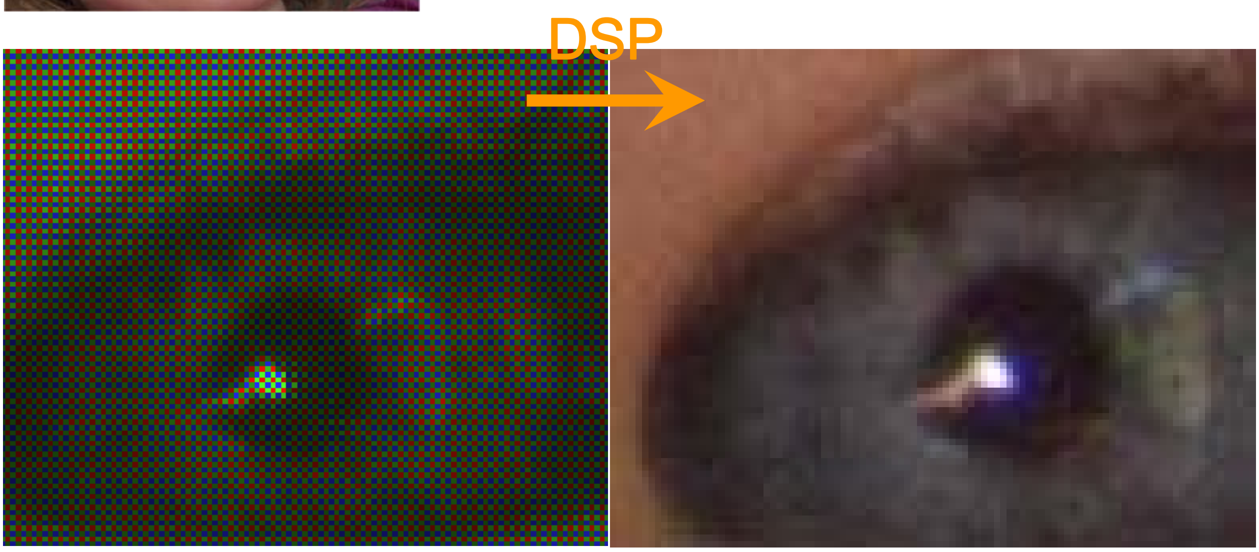
Example II: Digital Imaging Camera



Focus/exposure Control → Pre-processing → White Balancing → Demosaic → Color Transform → Post-processing → Compression

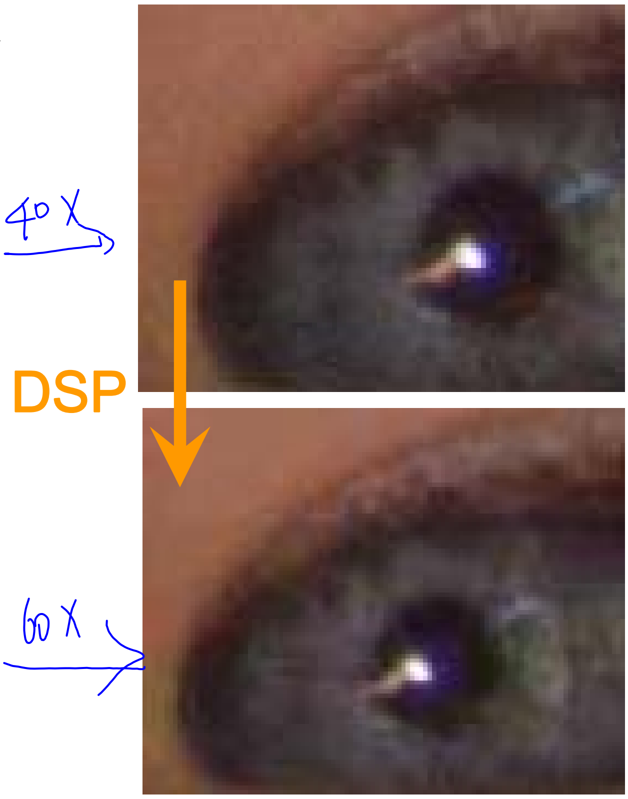
→去马赛克





- Compression of 40x without perceptual loss of quality.

40 x



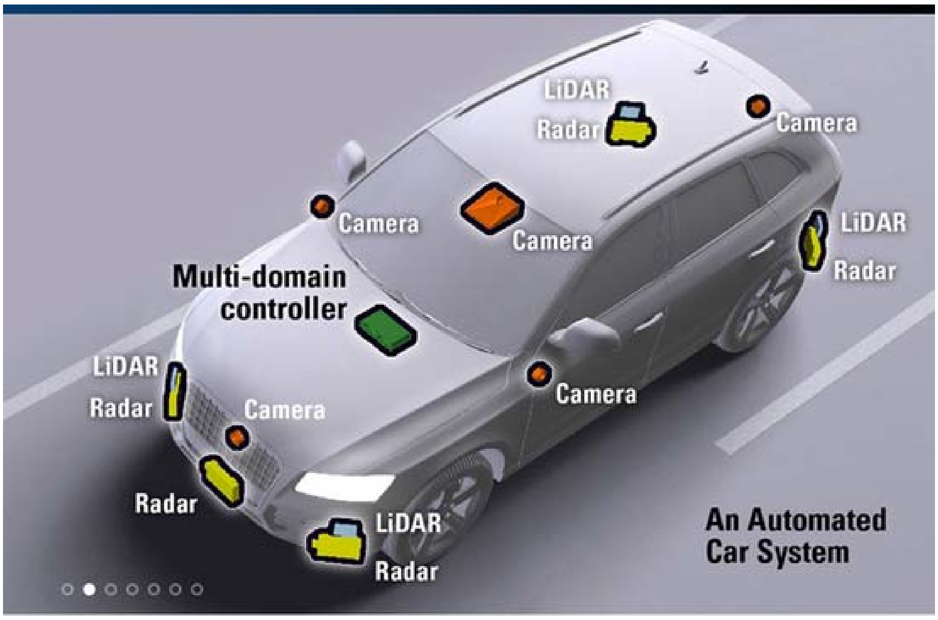
- Example of slight over compression:
  - difference enables x60 compression!

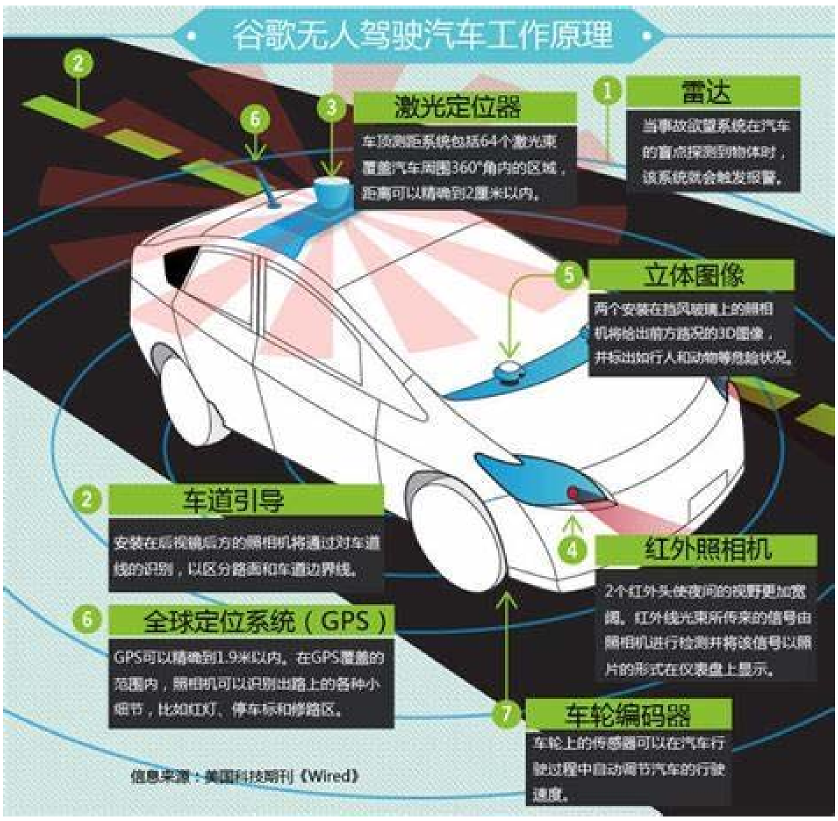
60 x

Computational Photography

Now implemented in smart phones (HDR) → high dynamic range

Example III Auto Drive





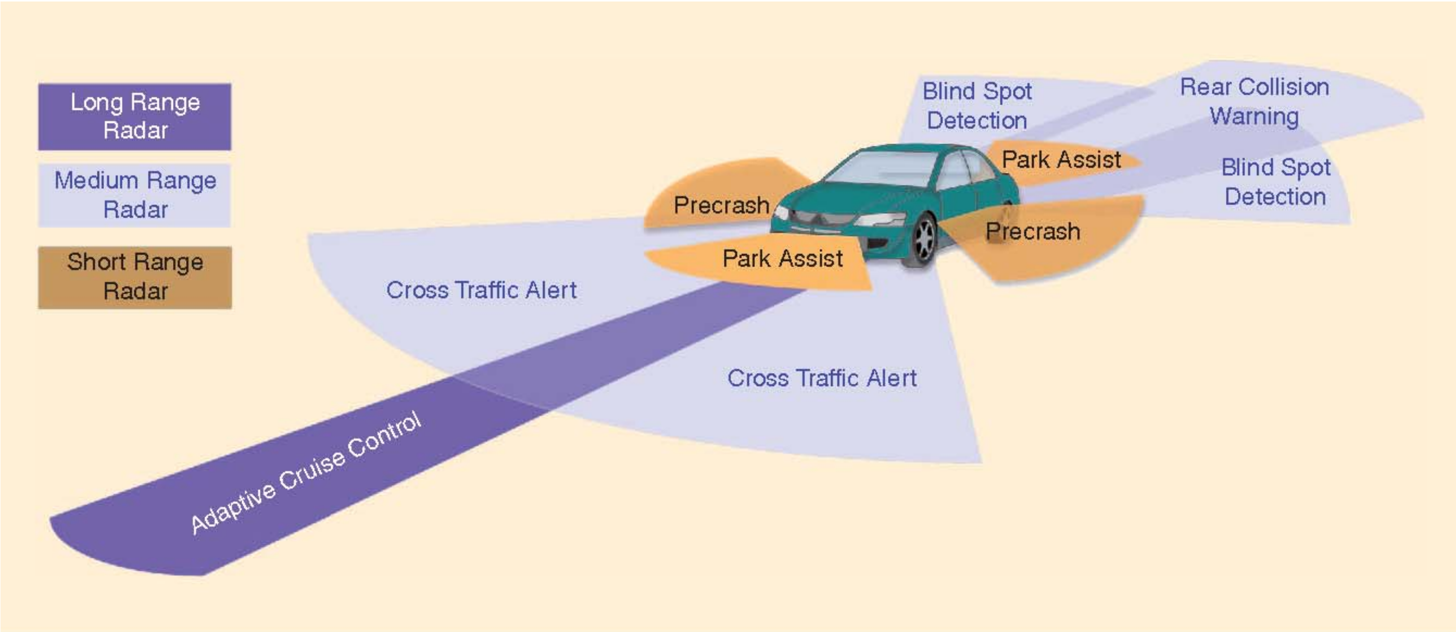


FIGURE 1. An ADAS consists of different range radars.